

**L_p-NORM ESTIMATION TECHNIQUES APPLIED
TO MULTIPLE EMITTER LOCATION**

By

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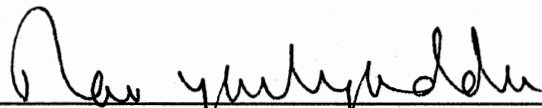
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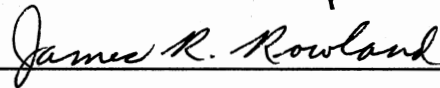
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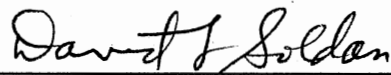
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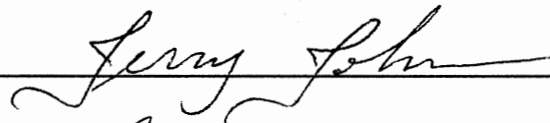
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PREFACE

An approach to L_p -norm based estimation of signal directions of arrival at an antenna array was developed. The Multiple Signal Classification (MUSIC) algorithm of R. O. Schmidt served as the base for this work. The approach is capable of resolving directions of arrival which MUSIC fails to isolate in the presence of impulsive noise. The approach is the first of its kind in explicitly considering values for p unequal to two.

Application of the L_p model to other beamforming algorithms is also discussed, and such application appears promising. Extension of the techniques described herein to other direction of arrival approaches is therefore encouraged.

I wish to express my sincere gratitude to all those who made it possible for me to conduct this research and prepare this dissertation. Although an exhaustive list would fill volumes, I can mention some of those most directly involved.

I am deeply indebted to my major adviser, Dr. Rao Yarlagadda, for his intelligent guidance, concern, and invaluable help. Dr. James Rowland deserves a special acknowledgement as well, both for serving on my committee and for providing the encouragement which convinced me to pursue a Ph.D. in the first place. No less thanks is due to the other members of my committee, Dr. Charles Bacon, Dr. Jerry Johnson, and Dr. Dave Soldan, for their patient assistance and advisement during this work. I would also like to thank all the other members of the Electrical Engineering faculty at Oklahoma State University, as

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LIST OF SYMBOLS

\mathbf{A}	– matrix \mathbf{A}
\mathbf{A}^T	– transpose of \mathbf{A}
\mathbf{A}^*	– complex conjugate of \mathbf{A}
\mathbf{A}^H	– complex conjugate transpose of \mathbf{A}
$\text{tr}\{\mathbf{A}\}$	– trace of matrix \mathbf{A}
$\text{rank}\{\mathbf{A}\}$	– rank of matrix \mathbf{A}
$ \mathbf{A} $	– determinant of matrix \mathbf{A}
$[a,b,\dots,c]$	– (column) vector with entries a, b, \dots, c
$\text{diag}(a,b,\dots,c)$	– diagonal matrix with entries a, b, \dots, c
\mathbf{f}	– (column) vector \mathbf{f}
$\mathbf{x} \cdot \mathbf{y}$	– inner ("dot") product of vectors \mathbf{x} and \mathbf{y} : $\mathbf{x} \cdot \mathbf{y} = \mathbf{x}^H \mathbf{y}$
$E\{\}$	– statistical expectation

CHAPTER I

INTRODUCTION

The problem of identifying the directions of arrival of signals from multiple emitters illuminating an antenna array is addressed. Use of the L_p norm, $1 \leq p \leq 3$, in estimating the arrival directions is considered. It is assumed that the emitters are in the far field with respect to the physical size of the sensor array, and that the received signals are narrow band. A signal is classified as narrow band if its bandwidth is small compared to the reciprocal of the transmission time of a wavefront across the array. Since the signals received by the sensors are produced by the emitters rather than the antenna elements, the direction finding system is considered passive rather than active.

The problem of direction of arrival estimation has been a topic of continuing interest to the engineering community for almost twenty years, having applications in areas as diverse as sonar, radar, radio-astronomy, seismology, and surveillance (Middleton, 1970) (Gallop, 1974) (Ouibrahim, 1986).

The Antenna Array Analysis Problem

Figure 1 shows a top view of an antenna array composed of N (six, in the case shown in the figure) receiving antennas, with M plane waves impinging on the array. Only one of the waves, the m^{th} , is shown for clarity of presentation. The system is to be used for azimuth-only direction of arrival determination (the antenna elements and wave propagation directions are coplanar). The antenna elements are located at positions r_1, r_2, \dots, r_N , and the M incident waves are

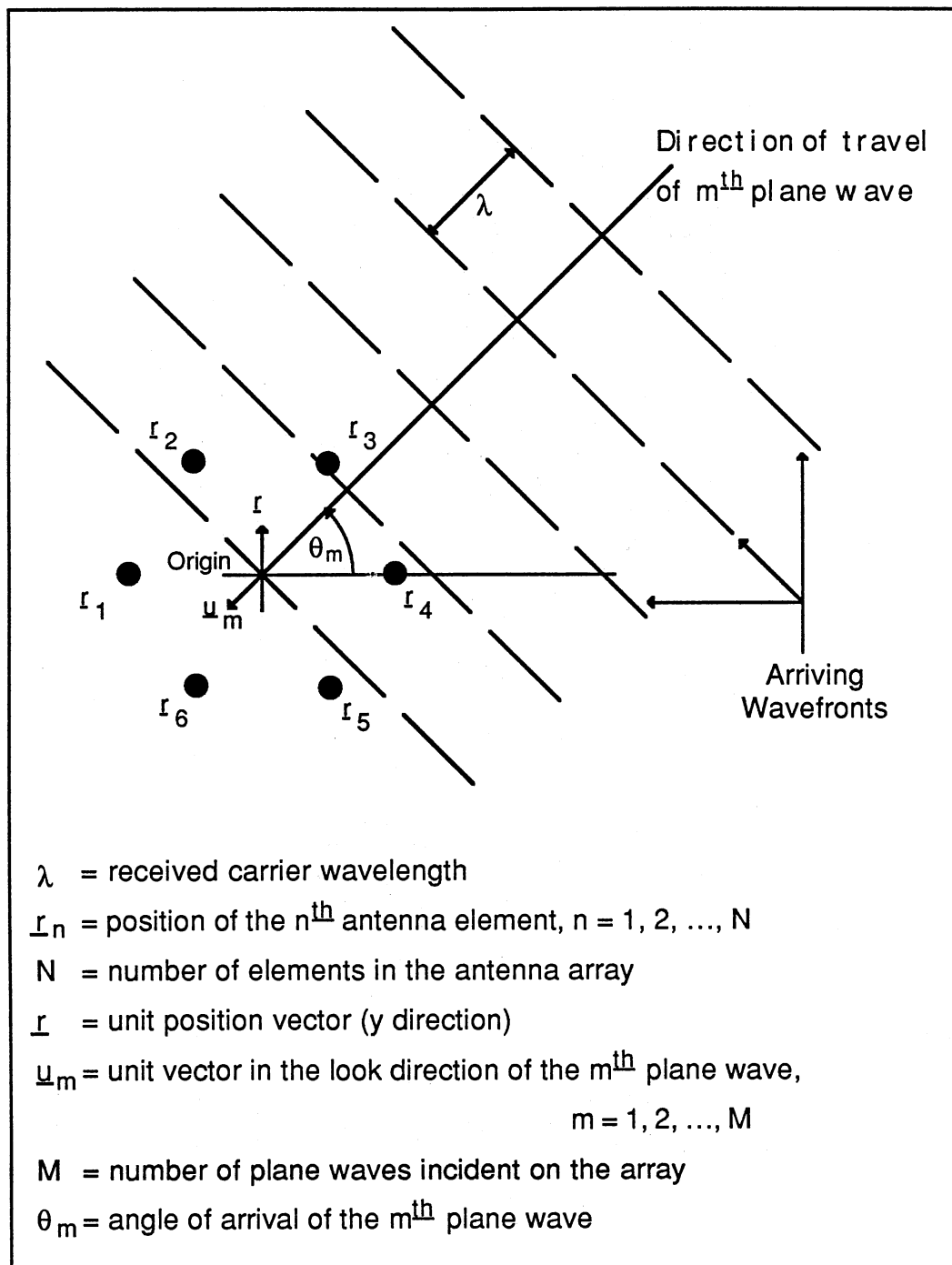


Figure 1. Geometry of an Antenna Array

impinging on the array at angles $\theta_1, \theta_2, \dots, \theta_M$ from the abscissa of the (arbitrarily chosen) coordinate system. All M waves are assumed to have the same wavelength, λ (the narrow band assumption).

If the received waves are modeled as complex sinusoids, then the real part of the noise-free signal received by element n due to received wave m at time t is given by:

$$\text{Re}\{s(n,m,t)\} = A_m \cos(2\pi f_c t + \psi_m(n) + \alpha_m) \quad (1-1)$$

where:

$s(n,m,t)$ = received signal as a function of n , m , and t .

n = array element number; $n = 1, 2, \dots, N$.

m = incident plane wave number; $m = 1, 2, \dots, M$.

t = time.

A_m = (real) amplitude of the m^{th} plane wave.

f_c = carrier frequency.

$\psi_m(n)$ = phase shift due to position of the n^{th} antenna element.

α_m = phase of $s(n,m,t)$ measured at the coordinate system origin.

The phase shift due to antenna element position is the projection of the distance vector from the origin to the element along the line of increasing phase of the incoming wave, and is expressed in Equation (1-2):

$$\psi_m(n) = 2\pi \frac{-\mathbf{r}_n \cdot \mathbf{u}_m}{\lambda} = -2\pi \frac{\mathbf{r}_n \cdot \mathbf{u}_m}{\lambda} \quad (1-2)$$

where \mathbf{u}_m is the unit vector in the look direction of the m^{th} plane wave, and $\mathbf{r}_n \cdot \mathbf{u}_m$ is the dot (inner) product of the vectors \mathbf{r}_n and \mathbf{u}_m . The negative sign in (1-2) arises from the fact that the direction of increasing phase of the incoming wave is opposite to the direction of \mathbf{u}_m shown in Figure 1. Incorporating the above expression into Equation (1-1), and expressing the result in phasor notation, gives:

$$s(n, m) = A_m e^{j\left(\frac{-2\pi \mathbf{r}_n \cdot \mathbf{u}_m}{\lambda} + \alpha_m\right)} \quad (1-3)$$

The noise-free signal received at each antenna element is the sum of the contributions from the M received waves:

$$s(n) = \sum_{m=1}^M A_m e^{-j\left(\frac{2\pi}{\lambda} \mathbf{r}_n \cdot \mathbf{u}_m\right)} e^{j\alpha_m} \quad (1-4)$$

$$\mathbf{r}_n \cdot \mathbf{u}_m = |\mathbf{r}_n| \cos \delta_{nm}; \quad \delta_{nm} = \text{angle between } \mathbf{r}_n \text{ and } \mathbf{u}_m$$

Rewriting Equation (1-4) in vector form gives:

$$\underline{\mathbf{s}} = \mathbf{A} \underline{\mathbf{f}} \quad (1-5)$$

where:

$$\underline{\mathbf{s}} \equiv \begin{bmatrix} s(1) \\ s(2) \\ \vdots \\ s(N) \end{bmatrix}; \quad \mathbf{A} \equiv \begin{bmatrix} e^{-j\frac{2\pi}{\lambda} \mathbf{r}_1 \cdot \mathbf{u}_1} & e^{-j\frac{2\pi}{\lambda} \mathbf{r}_1 \cdot \mathbf{u}_2} & \dots & e^{-j\frac{2\pi}{\lambda} \mathbf{r}_1 \cdot \mathbf{u}_M} \\ e^{-j\frac{2\pi}{\lambda} \mathbf{r}_2 \cdot \mathbf{u}_1} & e^{-j\frac{2\pi}{\lambda} \mathbf{r}_2 \cdot \mathbf{u}_2} & \dots & e^{-j\frac{2\pi}{\lambda} \mathbf{r}_2 \cdot \mathbf{u}_M} \\ \vdots & \vdots & \ddots & \vdots \\ e^{-j\frac{2\pi}{\lambda} \mathbf{r}_N \cdot \mathbf{u}_1} & e^{-j\frac{2\pi}{\lambda} \mathbf{r}_N \cdot \mathbf{u}_2} & \dots & e^{-j\frac{2\pi}{\lambda} \mathbf{r}_N \cdot \mathbf{u}_M} \end{bmatrix};$$

$$= [\underline{\mathbf{a}}_1 \quad \underline{\mathbf{a}}_2 \quad \dots \quad \underline{\mathbf{a}}_M]$$

$$\underline{\mathbf{f}} \equiv \begin{bmatrix} A_1 e^{j\alpha_1} \\ A_2 e^{j\alpha_2} \\ \vdots \\ A_M e^{j\alpha_M} \end{bmatrix}$$

The matrix \mathbf{A} contains a complete description of the effects of antenna array geometry on the received signal vector $\underline{\mathbf{s}}$. The vector $\underline{\mathbf{f}}$ completely describes the effects of the transmitted signal characteristics on the received signal vector $\underline{\mathbf{s}}$. The columns of \mathbf{A} , $\underline{\mathbf{a}}_m$, express the effect of array geometry on each

of the M individual incident waves. Plainly, the dot products $\mathbf{r}_n \cdot \mathbf{u}_m$ depend (albeit in a nonlinear fashion) on the incident wave arrival angles θ_m . Therefore, determining the entries in the vectors \mathbf{a}_m is tantamount to determining the angles of arrival of the incident plane waves. The vectors $\mathbf{a}_m = \mathbf{a}(\theta_m)$ are called the *signal subspace* vectors or just the *signal* vectors.

Although the matrix \mathbf{A} depends on θ_m , it of course also depends on antenna element range from the origin and on the received signal frequency. Additionally, although the relationship is not shown in Equation (1-5), the received signal vector is affected by other characteristics of the individual antennae (polarization, response pattern, etc.). Obviously, imperfect knowledge of these and other influences would result in errors in any attempt to determine emitter direction. Schmidt (1981) addresses the influence of polarization and other antenna characteristics on the matrix \mathbf{A} (the "array manifold" of the antenna array). Additional research is currently in progress on the effects of variations in received frequencies and errors in sensor location information (Shaw and Kumaresan, 1987) (Lo and Marple, 1987). Although these and other related issues are certainly important concerns, the research addressed herein centers around the emitter location process itself. It will therefore be assumed that satisfactory descriptions are available of all such "extraneous" influences on the signal vector (\underline{s}).

The total signal from the antenna array is the sum of the received signal and noise. Representing the noise at the various array elements by a noise vector, \underline{w} , the total signal observed by the antenna array is given by \underline{x} , as follows:

$$\underline{w} \equiv \begin{bmatrix} w(1) \\ w(2) \\ \vdots \\ w(N) \end{bmatrix}$$

$$\underline{x} = \underline{s} + \underline{w}$$

$$\underline{x} = \mathbf{A}\underline{f} + \underline{w} \quad (1-6)$$

It can be seen from the form of Equation (1-6) that the emitter location problem is in fact a problem of system identification. In an "ordinary" system identification problem, one is given or postulates a (typically) linear structure for the unknown system's dynamics. This takes the form of a vector equation such as:

$$\underline{x}_{n+1} = \Phi_n \underline{x}_n + \underline{b}_n \quad (1-7)$$

where \underline{x}_n is the system state, Φ_n is the state transition matrix, and \underline{b}_n is the input vector, at time t_n . The system identification problem consists of the attempt to determine the entries in Φ_n (i.e. the system parameters) by suitable analysis of sets of known or measured system states (\underline{x}_n) and inputs (\underline{b}_n).

In comparing Equations (1-6) and (1-7), it is clear that the antenna array problem is posed in a form very like that of the "standard" linear system formulation, with the sensor output vector acting as a "system state" (at time t_{n+1}), the \mathbf{A} matrix corresponding to the state transition matrix, and the noise vector corresponding to the system "input" vector. Thus, determination of wave arrival direction (i.e. the columns of \mathbf{A} in Equation (1-6)) using sensor output vector samples and known or assumed noise characteristics is completely analogous to the "normal" system identification problem, if the system to be identified is driven by a pure noise input. Indeed, the "output-input correlation" method of system identification is based on using a psuedo-random noise sequence as input to the system to be identified (Graupe, 1976) (Goodwin and Sin, 1984).

Thesis Summary

The remainder of this dissertation is organized as follows. Chapter II describes a number of historical approaches to the direction-of-arrival estimation problem. Techniques addressed are the Fourier method, the Maximum-Likelihood approach, the Adaptive Antenna method, and two methods based on linear prediction. The linear prediction approaches discussed are the Burg Technique and the Modified Forward-Backward Linear Predictor. The final traditional method addressed is the Multiple Signal Classification (MUSIC) algorithm. Chapter II closes with a discussion of L_p approaches to direction of arrival estimation.

The application of L_p -norm measures to the beamforming problem is discussed in detail in Chapter III. A development of the Iteratively Reweighted Least Squares (IRLS) algorithm is presented, followed by a discussion of methods for estimating the source covariance matrix. The approaches addressed are the enumeration method, the sum of powers method, and the L_1 selected rows method. Following this discussion, an expression for computing L_p estimates of directions of arrival (DOA) is proposed. Noise aspects are then addressed, followed by a discussion of adaptive approaches to the DOA problem.

Chapter IV presents the results of experiments conducted during this research, including DOA spectra for various values of p and the effects on DOA spectra of iterative DOA revision.

The conclusions drawn from the described research are discussed in Chapter V, which also includes a presentation of areas which show promise for further research in the L_p -DOA problem area.

CHAPTER II

PREVIOUS RESULTS

A wide variety of approaches to the emitter location problem has been employed in the past. Haykin (1985) provides an excellent summary of many of these techniques. The historical approaches discussed here are the Fourier method, the maximum-likelihood method, the adaptive antenna method, the Burg technique, the modified forward-backward linear predictor method, and the MUSIC algorithm.

Fourier Method

In the Fourier method, the "angle spectrum" of the received "spatial series" (the entries in the received signal vector, taken as a sequence in space) is computed. The angle spectrum is defined in a manner similar to that in which the Fourier spectrum of a time series is defined, as follows:

$$X(\phi) \equiv \sum_{n=1}^N x(n) e^{jn\phi} \quad (2-1)$$

$$\phi = \frac{2\pi d}{\lambda} \sin \theta$$

where $x(n)$ is an output sample from antenna number n , ϕ is the "electrical phase angle" for the array, d is the separation between array elements, λ is received signal wavelength, and θ is angle of arrival. The separation d , above, refers to elements in a colinear array, and therefore the above expression for electrical phase angle applies to colinear arrays only. The square of the

magnitude of the angle spectrum is called the "periodogram".

The essence of the Fourier method is to detect the maxima in a plot of the periodogram as a function of the electrical phase angle ϕ . The angles ϕ at which the periodogram attains a maximum are related to wave arrival angles, as can be seen intuitively by comparing Equation (2-1) to Equation (1-4). In a sense, one can say that the effect of forming the angle spectrum is to "remove" phase shifts due to the angle ϕ , so that contributions to the periodogram from the separate antennas add "in phase" when ϕ corresponds to a wave arrival angle.

Some of the advantages of the Fourier approach are that it is simple in concept, and that it is non-parametric. By "non-parametric", it is meant that the method does not depend on the formulation of a model for the process generating the received data (apart from the model of the receiving array). On the other hand, it provides relatively poor resolution, even when only a single incident wave is present. This is due to a comparatively wide main lobe and the presence of side lobes in the periodogram of a received wave. The poor resolution of the Fourier approach makes it inappropriate for use in many situations, particularly when multiple closely-spaced emitters are present. A related disadvantage of the Fourier method is that resolution can be improved only by increasing the array aperture, which in turn requires increasing the physical size of the array.

Maximum-Likelihood Method

This method involves postulating a model of the conditional probability density function for the received antenna output samples (Böhme, 1983) (Van Trees, 1968). The probability density function used is expressed in Equation (2-2), below:

$$f(\underline{x}_c | \underline{q}, \underline{\phi}) = K e^{\left[\frac{-S(\underline{x}_c)}{2\sigma^2} \right]} \quad (2-2)$$

$$\text{where } S(\underline{x}_c) = \sum_n \left\{ x_c(n) - \sum_{m=1}^{2M} q_m s_m(n) \right\}^2$$

In Equation (2-2), f is a conditional probability density function, K is a constant, \underline{x}_c is the "in-phase" component of the observed sample vector, and \underline{q} is a vector of values dependent on the incident waves' amplitudes, phases, and angles of arrival. The vectors $\underline{\phi}$ and \underline{s}_m are dependent only on the angles of arrival of the incident waves, and $s_m(n)$ is the n^{th} element of vector \underline{s}_m . The entries of $\underline{\phi}$ are the "electrical phase angles" of the various arriving waves, and are defined as in Equation (2-1). This implies that this development is valid for the colinear array case only. The vectors \underline{s}_m are obtained from a Gram-Schmidt orthonormalization procedure (Haykin, 1983) applied to a matrix of functions of the elements of $\underline{\phi}$. The value σ^2 is the variance of the (assumed Gaussian) noise at each antenna element, M is the number of incident waves, n is the antenna element number, and m is the incident plane wave number. The "in-phase" component of the sample vector is the real part of the complex sinusoids being received.

The probability density function, $f(\underline{x}_c | \underline{q}, \underline{\phi})$, is the probability of a particular value for the in-phase component of the observed sample vector, given selected values of the vectors \underline{q} and $\underline{\phi}$. The "particular value", given in Equation (2-3), is the value which would result, in the absence of noise, from M plane waves as described by the selected \underline{q} and $\underline{\phi}$ impinging on the array.

$$\begin{aligned}
x_c(n) = & \sum_{m=1}^M a_m (\cos \psi_m) \cos(n\phi_m) \\
& + \sum_{m=1}^M -a_m (\sin \psi_m) \sin(n\phi_m)
\end{aligned} \tag{2-3}$$

where:

$x_c(n)$, m , and M are as defined above,

ϕ_m is the electrical phase angle of the m^{th} plane wave
(the m^{th} element of ϕ),

a_m is the complex amplitude of the m^{th} plane wave, and

ψ_m is the phase of the m^{th} plane wave.

In the maximum likelihood approach, a likelihood function $f(\phi|x_c)$ is formed from the probability density function given in Equation (2-2). This likelihood function is simply the probability that a selected output sample vector results from a given electrical phase angle vector. To determine the actual electrical phase angle vector (and thus the actual angles of arrival of the incoming waves), the likelihood function is maximized. That is, initial values are guessed for ϕ_m , the likelihood function is evaluated based on the initial estimates, new estimates of ϕ_m are computed from the results, and the process is repeated until some iteration termination criterion is met.

The maximum likelihood approach shows resolution generally superior to that possible with the Fourier method, however, it suffers from several disadvantages. One is that, in order to remove the effects on the likelihood function of the vector \mathbf{q} (which embodies the transmitted waves' characteristics), it is necessary to assume a uniform joint probability density function for ϕ and \mathbf{q} , which may not be justifiable in practice. Also, the development above applies to co-linear arrays only, and a maximum likelihood approach for arbitrary array

geometry is likely to be considerably more complicated. Another disadvantage is computational complexity, since both a Gram-Schmidt orthonormalization procedure and a multi-variate maximization procedure must be performed at each iteration step (Fletcher and Powell, 1963) (Bard, 1974) (Bandler, 1973). This disadvantage is less important than it once would have been, since very sophisticated digital processing capabilities are now relatively inexpensive to obtain; however, computational complexity is still a consideration in selection of an approach to be used. A final disadvantage is that the surfaces over which the multi-variate maximization process must be performed are often ill-conditioned, resulting in numerical problems and possible loss of resolution.

Adaptive Antenna Method

This method applies directly only to the case of a colinear antenna array in a symmetric multipath environment (Kesler, 1981). When "symmetric multi-path" transmission is present, the array is receiving a single plane wave from a direction θ from the normal to the array, along with a single reflection of the wave from a direction $-\theta$ from the normal to the array.

In the adaptive antenna approach, a beam forming network and an adaptive canceller are used to minimize a weighted difference between a "reference beam" and two "auxiliary beams". The reference beam is produced by forming a weighted sum of the received antenna output samples, as shown in Equation (2-4):

$$X_R = \sum_{n=-\frac{N-1}{2}}^{\frac{N-1}{2}} b_n x_n \quad (2-4)$$

In Equation (2-4), X_R is the output of the reference beam, N is the number of array elements, n is the number of the array element under consideration, $\{b_n\}$ is a set of real weights to be applied to the antenna outputs, and x_n is the output of the n^{th} array element. The array elements are numbered such that element number 0 is in the center of the (colinear) array. The set of weights b_n are constrained to be symmetric ($b_{-n} = b_n$), and a binomially weighted array has been found to yield good overall performance (Haykin and Kesler, 1983).

The two auxiliary beams, X_1 and X_2 , are formed by summing weighted, phase-shifted samples from each of the array elements, as shown in Equation (2-5):

$$\left. \begin{aligned} X_1 &= \sum_{n=-\frac{N-1}{2}}^{\frac{N-1}{2}} b_n x_n e^{jn\phi} \\ X_2 &= \sum_{n=-\frac{N-1}{2}}^{\frac{N-1}{2}} b_n x_n e^{-jn\phi} \end{aligned} \right\} \quad (2-5)$$

The factor ϕ , by which the samples are successively phase shifted in the auxiliary beams, is the electrical phase angle of the direction (relative to the normal to the array) of the first auxiliary beam.

In the adaptive antenna technique, the reference beam acts as a "desired response" for an adaptive controller. The controller adjusts a real weight, w , which is multiplied by the sum of the two auxiliary beams. The controller is constructed to minimize the mean square value of the "error", which is defined as the difference between the "desired response" (reference beam) and the sum of the auxiliary beams times the real weight, w , described above. The antenna is considered "nulled" when the mean square error is minimized. It can be shown

that, when the adaptive antenna is nulled, the weight w is a direct function of the angle of arrival of the primary wave, which can be estimated from a graph of weight versus arrival angle.

The advantages to this technique are that it is comparatively straightforward in implementation and does not involve a large computational burden. Its primary disadvantage is that it is limited in application, being restricted to symmetric multipath environments and colinear arrays only. Another disadvantage is that it exhibits poor resolution for angles of arrival near the normal to the array.

Linear Predictive Methods

In the linear prediction approach, the output, x_n , from antenna element n is "predicted" as a linear combination of the outputs from the T "previous" elements in the array, as in Equation (2-6):

$$\hat{x}_n = \sum_{t=1}^T h_t x_{n-t}; \quad T < n \leq N \quad (2-6)$$

where h_t are the "prediction filter coefficients" and T (the number of spatial samples used in the "prediction") is called the "order" of the prediction filter.

The forward prediction error, or residual, is defined as the difference between the observed (measured) output of the n^{th} antenna element and its predicted value. The term "forward" is used because the "prediction" is made using samples that are considered to be "previous" (in space) to the n^{th} element. This is an arbitrary distinction, and a completely analogous "backward" prediction error will be introduced later. Evaluating the difference between the actual and predicted outputs yields:

$$f(T, n) = \sum_{t=0}^T r_t x_{n-t}; \quad T < n \leq N \quad (2-7)$$

where:

$f(T, n)$ = forward prediction error using a filter of order T at antenna element n , and

$$r_t = \left\{ \begin{array}{ll} 1; & t = 0 \\ -h_t; & t = 1, 2, \dots, T \end{array} \right\} \quad (2-8)$$

The spatial filter defined by the set of coefficients r_t is called a prediction error filter of order T . If the vectors \underline{r} and \underline{x} are defined as:

$$\underline{r} \equiv \begin{bmatrix} r_0 \\ r_1 \\ \vdots \\ r_T \end{bmatrix}; \quad \underline{x}(n) \equiv \begin{bmatrix} x_n \\ x_{n-1} \\ \vdots \\ x_{n-T} \end{bmatrix} \quad (2-9)$$

then the mean-square value of the forward prediction error is:

$$\epsilon_f \equiv E\{|f(T, n)|^2\} = \underline{r}^H \mathbf{R} \underline{r} \quad (2-10)$$

where $\mathbf{R} \equiv E\{\underline{x}^*(n) \underline{x}^T(n)\}$, $E\{\}$ denotes the expectation operator, $*$ denotes complex conjugate, T denotes transpose, and H denotes complex conjugate transpose. Defining a similar filter in the "backward" direction produces:

$$b(T, n) = \sum_{t=0}^T r_{bt} x_{n-t}; \quad T < n \leq N \quad (2-11)$$

$$r_{bt} = \left\{ \begin{array}{ll} -h_{bt}; & t = 0, 1, 2, \dots, T-1 \\ 1; & t = T \end{array} \right\} \quad (2-12)$$

$$\underline{r}_b \equiv \begin{bmatrix} r_{b0} \\ r_{b1} \\ \vdots \\ r_{bT} \end{bmatrix} \quad (2-13)$$

$$\varepsilon_b \equiv E\{ |b(T,n)|^2 \} = \underline{r}_b^H \mathbf{R} \underline{r}_b \quad (2-14)$$

The backward prediction error filter defined by the vector \underline{r}_b computes the difference between the observed value of x_{n-T} and its predicted value (since $r_{bT} = 1$).

It can be shown that when the average power of the prediction error is minimized, peaks will occur at points corresponding to the angles of arrival of the incident waves in the power spectrum of an all-pole filter constructed from the prediction coefficients. That is,

$$P(\phi) \equiv \frac{K}{\left| 1 + \sum_{t=1}^T r_t e^{-jt\phi} \right|^2}$$

where K is a constant and ϕ is the electrical phase angle (Equation (2-1)), exhibits peaks at values of ϕ corresponding to the directions of arrival of the incident waves. This expression uses only the forward error coefficients (r_t); it may be that a similar expression involving both the forward and backward coefficients would be of use as well.

The matrix \mathbf{R} is Hermitian positive semidefinite for any vector $\underline{x}(n)$, and is usually positive definite. Additionally, when the input vector $\underline{x}(n)$ is "spatially stationary", \mathbf{R} is Toeplitz as well. By "spatially stationary", it is meant that the statistics of the i th output with respect to the j th output depend only on the difference between i and j , not their actual values.

When \mathbf{R} is both Hermitian and Toeplitz, the minimum mean square error for both the forward and backward filters is obtained by setting:

$$\underline{d} \equiv \begin{bmatrix} 1 \\ 0 \\ \vdots \\ 0 \end{bmatrix}; \quad \underline{d}_b \equiv \begin{bmatrix} 0 \\ 0 \\ \vdots \\ 1 \end{bmatrix} \quad (2-15)$$

$$\mathbf{R} \underline{r} = \varepsilon_{f,\min} \underline{d} \quad (2-16)$$

$$\mathbf{R}r_b = \varepsilon_{b,\min} \mathbf{d}_b \quad (2-17)$$

$$r_{bt} = r_{T-t}^*; \quad t = 0, 1, \dots, T \quad (2-18)$$

In this case, it is also true that $\varepsilon_{f,\min} = \varepsilon_{b,\min} = \varepsilon_{\min}$.

Burg Technique

In the Burg technique, recursive relations are developed from Equations (2-15) through (2-18) for Toeplitz \mathbf{R} to permit computation of the parameters of a prediction-error filter of order $T+1$ given the parameters of a filter of order T (Burg, 1967) (Burg, 1975). The resulting relationships are:

$$r_t(T+1) = r_t(T) + \Gamma_{T+1} r_{T+1-t}^*(T); \quad t = 0, 1, \dots, T \quad (2-19)$$

$$\varepsilon_{\min}(T+1) = (1 - |\Gamma_{T+1}|^2) \varepsilon_{\min}(T) \quad (2-20)$$

where Γ_{T+1} is a coefficient to be computed at each increase in filter order.

Equations (2-19) and (2-20) are known as the Levinson-Durbin recursion (Levinson, 1947) (Durbin, 1960). Prediction errors at each order can be evaluated using the prediction errors from the previous order by:

$$f(T,n) = f(T-1,n) + \Gamma_T b(T-1,n-1) \quad (2-21)$$

$$b(T,n) = b(T-1,n-1) + \Gamma_T^* f(T-1,n) \quad (2-22)$$

Equations (2-21) and (2-22) give rise to a lattice filter structure, permitting comparatively simple transition from one filter order to the next. The value of Γ_T is computed by minimizing the sum of the mean square values of the forward and backward prediction errors for a filter of order T . That is, the quantity $J(T) \equiv E\{|f(T,n)|^2 + |b(T,n)|^2\}$ is minimized. This produces:

$$\Gamma_{\text{opt}, T} = - \frac{2E\{f(T-1,n)b^*(T-1,n-1)\}}{E\{|f(T-1,n)|^2 + |b(T-1,n-1)|^2\}} \quad (2-23)$$

Equation (2-23) is known as the Burg formula.

The Burg technique for emitter location starts with a filter of order 0, wherein $f(0,n) = b(0,n) = x_n$ for n from 1 to N and $\epsilon_{\min}(0)$ is the average of $|x_n|^2$. Filter order is then repeatedly incremented (computing \underline{r} and \underline{r}_b at each stage) until some desired threshold is reached in the mean-square error sum. The power spectrum of the all-pole filter constructed from the resulting prediction coefficients is then plotted, and the peaks in that spectrum are taken as the angles of arrival.

The advantages to the Burg technique are that it exhibits improved resolution over the Fourier method and is more generally applicable than the adaptive antenna approach. Its disadvantages are that most of the work with it has been applied to colinear antenna arrays only, and that the sample vector must be spatially stationary. The spatial stationarity requirement is violated in the case of arbitrarily coherent arriving waves, which makes the method unsuitable for a number of environments (e.g. specular multipath).

Modified Forward-Backward Linear Predictor

The (standard) forward-backward linear predictor (FBLP) approach is based on the same principles as the Burg technique (i.e. linear prediction) (Ulriych and Clayton, 1976) (Nuttal, 1976). However, the prediction error energy to be minimized in the FBLP approach is defined to be:

$$\epsilon \equiv \sum_{n=T+1}^N (|f(T,n)|^2 + |b(T,n)|^2) = \underline{r}^H \mathbf{C} \underline{r} \quad (2-24)$$

where:

$$\mathbf{C} \equiv \sum_{n=T+1}^N [\underline{x}^*(n)\underline{x}^T(n) + \underline{x}_b(n)\underline{x}_b^H(n)] \quad (2-25)$$

and the vectors \underline{r} and \underline{x} are defined in a manner analogous to that for the Burg technique. The matrix \mathbf{C} in Equation (2-25) is, in general, non-Toeplitz. The standard FBLP approach exhibits improved performance over the Burg technique, however, its performance is limited by the noise contributions to any estimate of the matrix \mathbf{C} .

In the modified forward-backward linear predictor (MFBLP) method, the estimate of \mathbf{C} is replaced by an estimate \mathbf{D} , which is of the same dimensions but of lower rank (Kumaresan and Tufts, 1981) (Tufts and Kumaresan, 1982) (Eckart and Young, 1931). The matrix estimate \mathbf{D} is constructed using selected eigenvectors of a submatrix of the estimate of \mathbf{C} . By discarding eigenvectors corresponding to "small" eigenvalues (which are introduced by observation noise), the selection process provides a means to effectively increase signal to noise ratio, permitting an increase in filter order and a corresponding increase in resolution.

The advantages of the modified forward-backward linear predictor approach are that it exhibits increased resolution over the Burg technique and that it works well even in the presence of arbitrarily coherent arriving waves (since it does not require that the covariance matrix be Toeplitz). Its primary disadvantage is that most of the work using it has been applied to colinear antenna arrays only.

The MUSIC Algorithm

The MUSIC (Multiple Signal Classification) algorithm of R. O. Schmidt (1981) employs an eigenvector decomposition of the sample autocovariance matrix estimate to develop a function with peaks at the angles of arrival of the incident waves (Speiser, 1985). The technique is also applicable to estimation of the frequencies of sinusoids present in noise. Eigenvector techniques have

shown improved performance over spectral approaches to this problem, such as autoregressive or Prony (1795), particularly with lower signal to noise ratios and in the presence of closely-spaced narrow band spectral components (Hildebrand, 1956). Development of the MUSIC approach to emitter location is discussed in this section.

Recalling Equation (1-6), it can be seen that a sequence of samples \underline{w}_s of the (time-varying) noise vector $\underline{w}(t)$ can be considered to be the output of a random process. Thus, samples \underline{x}_s of the total signal vector $\underline{x}(t)$ can also be viewed as the output of a random process (actually the sum of a deterministic process and a purely random process). The covariance matrix associated with the total signal vector is (where $E\{\}$ signifies expectation over s):

$$\mathbf{R}_x \equiv E\{\underline{x}\underline{x}^H\} = E\{(\mathbf{A}\underline{f} + \underline{w})(\mathbf{A}\underline{f} + \underline{w})^H\}$$

Assuming that the noise (\underline{w}) is zero mean and independent of the incoming signals ($\mathbf{A}\underline{f}$):

$$\mathbf{R}_x = \mathbf{A}E\{\underline{f}\underline{f}^H\}\mathbf{A}^H + E\{\underline{w}\underline{w}^H\} = \mathbf{R}_s + \mathbf{R}_w$$

where:

$$\mathbf{R}_s \equiv \mathbf{A}\mathbf{P}\mathbf{A}^H; \quad \mathbf{P} \equiv E\{\underline{f}\underline{f}^H\}; \quad \mathbf{R}_w \equiv E\{\underline{w}\underline{w}^H\}$$

Now \mathbf{R}_x , \mathbf{R}_s , and \mathbf{R}_w are $N \times N$ matrices (where N is the number of antennas), and are non-negative definite based on their form. \mathbf{P} is an $M \times M$ matrix (where M is the number of incident waves) of rank M , and is, in general, positive definite. \mathbf{R}_w is also in general positive definite. The case where the number of arriving waves is less than the number of antennas ($M < N$) is of primary concern. Since \mathbf{P} is of rank M and \mathbf{A} is of rank $N > M$, \mathbf{R}_s is of rank M . \mathbf{R}_x and \mathbf{R}_w are assumed to be full rank (rank N).

If \mathbf{R}_w is rewritten as $\lambda\mathbf{R}_b$ for $\text{tr}\{\mathbf{R}_b\} = \text{rank}\{\mathbf{R}_b\} = N$ and some parameter λ , then \mathbf{R}_x is expressed by:

$$\mathbf{R}_x = \mathbf{R}_s + \lambda \mathbf{R}_b \quad (2 - 26)$$

$$\mathbf{R}_x - \lambda \mathbf{R}_b = \mathbf{R}_s$$

Since \mathbf{R}_s is singular (being of dimension N and rank $M < N$):

$$|\mathbf{R}_s| = |\mathbf{R}_x - \lambda \mathbf{R}_b| = 0$$

Plainly, the above equation can be true only for λ a generalized eigenvalue of the matrix pair $(\mathbf{R}_x, \mathbf{R}_b)$. That is, λ must be one of the solutions to Equation (2-27):

$$\mathbf{R}_x \mathbf{e}_n = \lambda_n \mathbf{R}_b \mathbf{e}_n; \quad n = 1, 2, \dots, N \quad (2 - 27)$$

It is clear from their definitions that \mathbf{R}_s , \mathbf{R}_x , and \mathbf{R}_b are Hermitian. It can be shown that, for Hermitian \mathbf{R}_x and \mathbf{R}_b , with either \mathbf{R}_x or \mathbf{R}_b or both positive definite, and any non-zero vector \mathbf{z} , there exists a non-singular transformation matrix \mathbf{V} such that, when $\mathbf{r} = \mathbf{V}\mathbf{z}$:

$$\mathbf{r}^H \mathbf{R}_x \mathbf{r} = \sum_{n=1}^N \lambda_n |z_n|^2, \quad \text{and} \quad \mathbf{r}^H \mathbf{R}_b \mathbf{r} = \sum_{n=1}^N |z_n|^2$$

where the λ_n satisfy Equation (2-27) (Hohn, 1964). A similar result holds for the "normal" eigenvalues of \mathbf{R}_x . That is, for any non-zero vector \mathbf{y} , there exists a non-singular transformation matrix \mathbf{U} such that, when $\mathbf{r} = \mathbf{U}\mathbf{y}$:

$$\mathbf{r}^H \mathbf{R}_x \mathbf{r} = \sum_{n=1}^N \mu_n |y_n|^2$$

where the μ_n satisfy $|\mathbf{R}_x - \mu_n \mathbf{I}| = 0$. Further, since \mathbf{R}_x is non-negative definite, $\mu_n \geq 0$, all n (> 0 for positive definite \mathbf{R}_x). Sylvester's Law of Inertia states that, regardless of the non-singular transformation used to reduce a Hermitian form to the form:

$$g_1 |z_1|^2 + g_2 |z_2|^2 + \dots + g_p |z_p|^2 - g_{p+1} |z_{p+1}|^2 - \dots - g_r |z_r|^2,$$

where the g_i 's are all positive, the integers p (the "index" of the matrix) and r (the rank of the matrix) are unchanged. That is, p and r do not change as a result of

using a different non-singular transformation. Thus, since $\mu_n \geq 0$, it follows that $\lambda_n \geq 0$ for all n .

Recalling Equations (2-26) and (2-27), we have:

$$\mathbf{R}_s \underline{e}_n + \lambda \mathbf{R}_b \underline{e}_n = \mathbf{R}_x \underline{e}_n = \lambda_n \mathbf{R}_b \underline{e}_n$$

or:

$$\mathbf{R}_s \underline{e}_n = (\lambda_n - \lambda) \mathbf{R}_b \underline{e}_n$$

That is, the eigenvalues of the matrix \mathbf{R}_s in the metric of \mathbf{R}_b differ from the eigenvalues of the matrix \mathbf{R}_x in the metric of \mathbf{R}_b (the λ_n) by λ (from Equation (2-26)) in every case. Since \mathbf{R}_s is non-negative definite, the quantities $\lambda_n - \lambda$ must all be ≥ 0 (using an inertia argument similar to that given above). Therefore, λ in Equation (2-26) must be the minimum of the generalized eigenvalues. Equation (2-26) can thus be rewritten as:

$$\mathbf{R}_x = \mathbf{R}_s + \lambda_{\min} \mathbf{R}_b \quad (2 - 28)$$

Now \mathbf{R}_s is singular with a zero (normal) eigenvalue of multiplicity K ($= N - M$). Therefore, the same argument which shows that $\lambda_n \geq 0$ implies that the minimum generalized eigenvalue of \mathbf{R}_x in the metric of \mathbf{R}_b (λ_{\min}) must also have multiplicity K . That is, $\lambda_n - \lambda_{\min}$ must be 0 for precisely K of the λ_n 's.

Since:

$$\mathbf{R}_s = \mathbf{A} \mathbf{P} \mathbf{A}^H$$

we have:

$$\mathbf{A} \mathbf{P} \mathbf{A}^H \underline{e}_n = (\lambda_n - \lambda_{\min}) \mathbf{R}_b \underline{e}_n$$

Arranging the eigenvalues in descending order gives

$$\lambda_1 \geq \lambda_2 \geq \dots > \lambda_{\min, K} = \lambda_{\min, K-1} = \dots = \lambda_{\min, 1} = \lambda_{\min}.$$

Thus $\mathbf{A} \mathbf{P} \mathbf{A}^H \underline{e}_n = \underline{0}$ for the K eigenvectors associated with λ_{\min} , and therefore $\underline{e}_n^H \mathbf{A} \mathbf{P} \mathbf{A}^H \underline{e}_n = (\mathbf{A}^H \underline{e}_n)^H \mathbf{P} (\mathbf{A}^H \underline{e}_n) = 0$. Since \mathbf{P} is positive definite, this in turn implies that:

$$\mathbf{A}^H \underline{e}_n = \underline{0} \text{ for the } K \text{ eigenvectors } (\underline{e}_n) \text{ associated with } \lambda_{\min}.$$

The K eigenvectors associated with λ_{\min} are called the "noise subspace" eigenvectors (or just the noise eigenvectors), and the remaining eigenvectors (associated with the non-minimum generalized eigenvalues) are called the "principal" eigenvectors (Marple, 1987). It has just been shown, then, that the noise eigenvectors are orthogonal to the signal vectors (the columns of \mathbf{A}). That is:

$$\underline{e}^H \underline{a}(\theta) = \underline{a}^H(\theta) \underline{e} = 0; \quad \underline{e} = \underline{e}_{M+1}, \underline{e}_{M+2}, \dots, \underline{e}_N; \quad \theta = \theta_1, \theta_2, \dots, \theta_M$$

The MUSIC algorithm uses a direction estimator, $f(\theta)$, formed from the noise eigenvectors as follows:

$$\mathbf{E} \equiv [\underline{e}_{M+1} \mid \underline{e}_{M+2} \mid \dots \mid \underline{e}_N]$$

$$g(\theta) = \underline{a}^H(\theta) \mathbf{E} \mathbf{E}^H \underline{a}(\theta)$$

$$f(\theta) = \frac{1}{g(\theta)} = \frac{1}{\underline{a}^H(\theta) \mathbf{E} \mathbf{E}^H \underline{a}(\theta)} \quad (2 - 29)$$

Since:

$$g(\theta) = \sum_{n=M+1}^N [\underline{a}^H(\theta) \underline{e}_n] [\underline{e}_n^H \underline{a}(\theta)] = \sum_{n=M+1}^N [\underline{a}^H(\theta) \underline{e}_n] [\underline{a}^H(\theta) \underline{e}_n]^*$$

it is clear that (theoretically) $g(\theta) = 0$ and therefore $f(\theta)$ is infinite for values of θ corresponding to the directions of arrival of the M incident waves ($\theta = \theta_1, \theta_2, \dots, \theta_M$). In practice, $f(\theta)$ as defined in Equation (2-29) exhibits sharp peaks at the arrival directions.

To summarize, the steps in the MUSIC algorithm for angle-of-arrival estimation are:

- 1) Collect sample vectors \underline{x}
- 2) Estimate $\mathbf{R}_x = E\{\underline{x}\underline{x}^H\}$ and $\mathbf{R}_w \equiv E\{\underline{w}\underline{w}^H\} = \lambda_{\min}\mathbf{R}_b$
- 3) Solve the generalized eigenproblem: $\mathbf{R}_x\underline{e} = \lambda\mathbf{R}_b\underline{e}$
- 4) Form an estimate, M' , of the number of arriving plane waves, M , by subtracting the multiplicity of the minimum eigenvalue, λ_{\min} , from the number of array elements, N
- 5) Form \mathbf{E} from $\underline{e}_{M'+1}, \underline{e}_{M'+2}, \dots, \underline{e}_N$
- 6) Evaluate $f(\theta) = \{\underline{a}^H(\theta)\mathbf{E}\mathbf{E}^H\underline{a}(\theta)\}^{-1}$, and identify M' peaks at $\theta_1, \theta_2, \dots, \theta_{M'}$.
- 7) Form an estimate, \mathbf{A}' , of \mathbf{A} as follows:

$$\mathbf{A}' = [\underline{a}(\theta_1) \mid \underline{a}(\theta_2) \mid \dots \mid \underline{a}(\theta_{M'})]$$

- 8) Estimate the source covariance matrix:

$$\mathbf{P}' = (\mathbf{A}'^H\mathbf{A}')^{-1}\mathbf{A}'^H(\mathbf{R}_x - \lambda_{\min}\mathbf{R}_b)\mathbf{A}'(\mathbf{A}'^H\mathbf{A}')^{-1}$$

The principal advantage of the MUSIC algorithm is that it is the most generally applicable high-resolution approach, applying as it does to completely arbitrary array geometry and functioning well in the presence of coherent arriving signals. Its disadvantages are that it requires a great deal of computation, and that its development is not straightforward (although its implementation is, at least conceptually).

For some array geometries, it is possible to drastically reduce the amount of computation required while retaining the high resolution of the MUSIC algorithm. The ESPRIT method of Paulraj, Roy, and Kailath can be applied to arrays consisting of matching element pairs, and does not require a search to determine directions of arrival (Speiser, 1987). ESPRIT is expected to perform as well as or better than MUSIC, its primary disadvantage being the loss of generality concerning array element placement.

L_p Methods

In the approaches described so far, estimates of directions of arrival are based on the L_2 norm; that is, the estimates produced are best in a "least squares" sense (although the quantities whose squares are minimized are different for each method). This dependence on least squares is explicit in some of the approaches (Burg, MFBLP, Maximum Likelihood), and only implicit in others (MUSIC, Fourier). When the noise present is Gaussian, and the system under study is linear, least squares estimates are also maximum-likelihood estimates. Thus, in the linear/Gaussian environment, there is little point in pursuing other than least squares approaches. However, the beamforming problem involves a highly nonlinear model (1-1). Also, it may not be reasonable to suppose that the noise to which the antenna array is subjected is invariably Gaussian. Geary (1947) states, "All texts should state: Normality is a myth, there has never been, and never will be, a normal distribution." Thus, least squares solutions may not be the ideal for the beamforming problem (at least not in all cases).

Yarlagadda, et al. (1985) point out that the L_p approach, where $1 \leq p < 2$, is less sensitive to aberrant noise than the L_2 approach. That is, minimizing a sum of error terms raised to some power (p) between 1 and 2 can produce a more robust estimator than one based on a minimum sum of squares. For example, since L_2 emphasizes all data equally, while L_1 is robust against outliers, L_1 generally produces superior results for impulsive noise. L_p measures can be used effectively to match the signal in a statistical sense. That is, if the density function has the form:

$$e^{-|r|^p}$$

i.e. a p -Gaussian function, then L_p methods give maximum likelihood estimates. For example, $p = 1$ corresponds to the Laplacian distribution and the L_1 measure gives maximum likelihood estimates for this case.

The L_p approach has been applied in a variety of areas including speech processing, seismic analysis, and geophysics (Denoel and Solvay, 1985) (Taylor, et. al., 1979) (Claerbout and Muir, 1973). On the other hand, very little work has been done in applying L_p methods to the direction of arrival problem. In fact, all the traditional approaches to the beamforming problem have been oriented toward obtaining L_2 solutions. The research described herein involves the use of the L_p , $1 \leq p \leq 3$, approach to estimation of directions of arrival and the source covariance matrix, \mathbf{P} .

CHAPTER III

L_p -NORM ESTIMATION APPROACHES

The estimate of \mathbf{P} given by step 8 in Chapter II, above, is a least-squares (L_2) estimate. The direction estimators described in Chapter II (including MUSIC) are L_2 estimators as well. As discussed in Chapter II, the research addressed herein applies the L_p norm to the direction of arrival problem.

Due to its ability to accommodate arbitrary array geometries, the MUSIC algorithm was chosen as the starting point for application of L_p norm based techniques. A number of L_p norm estimation algorithms exist (some of which are discussed in Chapter V), the most general being the Iteratively Reweighted Least Squares (IRLS) method.

Byrd and Pyne (1979) develop global convergence properties of the IRLS algorithm as applied to estimating a vector \mathbf{f} given the linear model:

$$\mathbf{x} = \mathbf{A}\mathbf{f} + \mathbf{w};$$

where \mathbf{x} is an N -element output vector, \mathbf{A} is an $N \times M$ matrix of rank M , \mathbf{f} is an M -element vector of unknown parameters, and \mathbf{w} is an N -element zero-mean noise vector. The IRLS algorithm repeatedly evaluates:

$$\mathbf{f}_{t+1} = [\mathbf{A}^H \mathbf{W}(\mathbf{f}_t) \mathbf{A}]^{-1} \mathbf{A}^H \mathbf{W}(\mathbf{f}_t) \mathbf{x} \quad (3-1)$$

until \mathbf{f} converges. The weighting matrix \mathbf{W} depends on the loss function to be minimized by the estimation. If $\mathbf{W} = \mathbf{I}$, Equation (3-1) clearly produces the L_2 estimate of \mathbf{f} ; i.e. the loss function minimized is the sum of the squares of the estimation residuals.

In the general form of the IRLS algorithm, \mathbf{W} is a diagonal matrix with entries:

$$w_{nn} = w_n = w(r_n) = \psi(r_n)/r_n = \rho'(r_n)/r_n \quad (3-2)$$

In Equation (3-2), $r_n = \mathbf{a}_n \mathbf{f} - x_n$; \mathbf{a}_n is the n^{th} row of \mathbf{A} ; x_n is the n^{th} element of \mathbf{x} ; and the objective function ρ is a positive, even, differentiable function. This definition of \mathbf{W} results in minimization of the loss function J , as defined below:

$$J = \sum_{n=1}^N \rho(\mathbf{a}_n \mathbf{f} - x_n) \quad (3-3)$$

That iteration of Equation (3-1) using \mathbf{W} as defined in Equation (3-2) minimizes J , above, is shown in Appendix A. Global convergence of the IRLS algorithm is guaranteed for objective functions ρ such that:

- (1) $\rho(r)$ is a differentiable, symmetric, positive function nondecreasing in $|r|$,
- (2) $\rho(r) \rightarrow \infty$ as $|r| \rightarrow \infty$,
- (3) $w(r) = \rho'(r)/r$ is nonincreasing in $|r|$, and
- (4) $w(r) = \rho'(r)/r$ is bounded for all r .

Application to the L_p minimization problem implies the use of the objective function $\rho(r) = |r|^p$. This function, however, gives rise to unbounded $w(r)$ as $r \rightarrow 0$ (violating condition 4). Huber and Dutter (1974) suggest the function:

$$\begin{aligned} w(r) &= |r|^{p-2}; \quad |r| > \varepsilon \\ &= \varepsilon^{p-2}; \quad |r| \leq \varepsilon \end{aligned} \quad (3-4)$$

for some small positive number ε . This function has been shown to give results which differ only negligibly from exact L_p solutions. Initialization of the IRLS algorithm is generally accomplished by starting with the L_2 estimate of \mathbf{f} . However, since the loss function J , and not the initial estimate, determines the nature of the estimate produced, other initial values of \mathbf{f} could reasonably be used.

The weight function (3-4) meets the convergence conditions for values of p between 1 and 2, thereby assuring global convergence of the IRLS algorithm to (approximate) L_p solutions for $1 \leq p \leq 2$. Additionally, local convergence of the IRLS algorithm using the function (3-4) is guaranteed for values of p up to 3 (Fletcher, et. al., 1971). Of course, since only local convergence is guaranteed for this case, the freedom of choice of initial $\hat{\mathbf{f}}$ may be lost.

It may be that L_p estimates with $p > 2$ are of use in some situations; however, such estimates would emphasize outliers more than L_2 estimates, so that the resulting estimators are likely to lack robustness. For $p < 1$, the problem space is not a normed linear space (the triangle inequality does not apply), so that the analysis breaks down. So, the emphasis during this research has been on p values between 1 and 2, with some effort directed toward values between 2 and 3.

L_p Estimates of the Source Covariance Matrix

As mentioned in Chapter II, one objective of direction of arrival (DOA) estimation is to produce an estimate of the source covariance matrix, \mathbf{P} . At least two distinct approaches to this problem are possible. The first involves estimating the signal vector samples and using those estimates to estimate \mathbf{P} . The second approach is to attempt to estimate \mathbf{P} directly from the problem parameters, bypassing estimation of the signal vector samples.

The second approach has the advantage of potentially requiring fewer computations, while the first approach leads to an algorithm for which Byrd and Pyne's (1979) analysis guarantees global convergence.

Enumeration Method

Given the model $\mathbf{x} = \mathbf{A}\mathbf{f} + \mathbf{w}$ from Chapter I, the L_p estimate of the signal vector sample (\mathbf{f}_s) which gives rise to the s^{th} array output sample (x_s) is (from (3-1) above):

$$\mathbf{f}_s = [\mathbf{A}^H \mathbf{W}(\mathbf{f}_s) \mathbf{A}]^{-1} \mathbf{A}^H \mathbf{W}(\mathbf{f}_s) x_s \quad (3-5)$$

This expression is correct, of course, only when IRLS convergence has been achieved (the value of \mathbf{f}_s has stopped changing).

The "enumeration" method uses the L_p estimates of all the signal vector samples to form an estimate of \mathbf{P} . Recalling the definition of \mathbf{P} , we may write:

$$\mathbf{P} \equiv E\{ \mathbf{f} \mathbf{f}^H \}$$

$$\hat{\mathbf{P}} = \frac{1}{S} \sum_{s=1}^S \mathbf{f}_s \mathbf{f}_s^H \quad (3-6)$$

where S total samples are used to form the estimate.

One potential disadvantage of this estimation approach is that, since neither the estimates \mathbf{f}_s nor the summation in (3-6) involves noise terms directly, no use is made of the known or assumed noise characteristics. Of course, this may not always be a problem, since no information about the noise characteristics may be available in a particular application. On the other hand, the "sum of powers" method, below, does take advantage of any available knowledge of noise parameters.

Sum of Powers Method

The estimate (3-1) can be improved by removing the noise from the observation vector, if the noise is known, as follows:

$$\mathbf{f}_\infty = [\mathbf{A}^H \mathbf{W}_\infty \mathbf{A}]^{-1} \mathbf{A}^H \mathbf{W}_\infty (\mathbf{x} - \mathbf{w}) \quad (3-7)$$

where the ∞ subscript denotes the values produced upon IRLS convergence. Substituting (3-7) into the definition of \mathbf{P} , and assuming that \mathbf{W} is Hermitian and will be evaluated in a way that does not depend on \mathbf{f} , one obtains:

$$\begin{aligned}
 \mathbf{P} &= E\{ \mathbf{f} \mathbf{f}^H \} \\
 &= E\{ [\mathbf{A}^H \mathbf{W}_\infty \mathbf{A}]^{-1} \mathbf{A}^H \mathbf{W}_\infty (\mathbf{x} - \mathbf{w})(\mathbf{x} - \mathbf{w})^H \mathbf{W}_\infty \mathbf{A} [\mathbf{A}^H \mathbf{W}_\infty \mathbf{A}]^{-1} \} \\
 &= [\mathbf{A}^H \mathbf{W}_\infty \mathbf{A}]^{-1} \mathbf{A}^H \mathbf{W}_\infty E\{ (\mathbf{x} - \mathbf{w})(\mathbf{x} - \mathbf{w})^H \} \mathbf{W}_\infty \mathbf{A} [\mathbf{A}^H \mathbf{W}_\infty \mathbf{A}]^{-1} \\
 \mathbf{P} &= [\mathbf{A}^H \mathbf{W}_\infty \mathbf{A}]^{-1} \mathbf{A}^H \mathbf{W}_\infty (\mathbf{R}_x - \lambda_{\min} \mathbf{R}_b) \mathbf{W}_\infty \mathbf{A} [\mathbf{A}^H \mathbf{W}_\infty \mathbf{A}]^{-1} \quad (3-8)
 \end{aligned}$$

Based on the form of (3-8), an iterative procedure for estimating \mathbf{P} can be proposed that is similar to the iteration of (3-1):

$$\mathbf{P}_{t+1} = (\mathbf{A}^H \mathbf{W}_t \mathbf{A})^{-1} \mathbf{A}^H \mathbf{W}_t (\mathbf{R}_x - \lambda_{\min} \mathbf{R}_b) \mathbf{W}_t \mathbf{A} (\mathbf{A}^H \mathbf{W}_t \mathbf{A})^{-1} \quad (3-9)$$

where \mathbf{W}_t is a weight matrix to be determined from the minimization process, and Equation (3-9) is iterated (starting with the L_2 solution) until some convergence criterion is satisfied.

For the above approach to be valid, the weight matrix \mathbf{W}_t must be Hermitian and must not depend directly on the individual samples of the source signal vector, \mathbf{f}_s . On the other hand, it is clear that estimates of the individual source signal vector samples must be computed based in some way upon \mathbf{W}_t , since such estimates are the only source of residuals to be used in computing the weight matrix. Given these constraints, a reasonable approach is to make \mathbf{W}_t a diagonal matrix based on a loss function formed from the sum (across samples) of estimation residuals raised to the p^{th} power:

$$r_n = \left\{ \sum_{s=1}^S | \mathbf{a}_n \mathbf{f}_s - x_{n,s} |^p \right\}^{\frac{1}{p}} \quad (3-10)$$

$$\rho(r_n) = | r_n |^p$$

$$\begin{aligned}
 w_n(r_n) &= \frac{\rho'(r_n)}{r_n} = p \frac{r_n^{p-1}}{r_n} = p r_n^{p-2} \\
 &= p \left\{ \sqrt[p]{\sum_{s=1}^S |a_{n,s}^f - x_{n,s}|^p} \right\}^{p-2} \quad (3-11)
 \end{aligned}$$

where $x_{n,s}$ is the n^{th} element of the s^{th} array output sample. In practice, the multiplicative constant p can be ignored in evaluating w_n , since it is common to all elements of \mathbf{W} and therefore cancels out in (3-1). Further, convergence is generally improved by normalizing the weights to a maximum value of 1. Additionally, since only the final value of \mathbf{W} is needed in Equation (3-8), the iteration process need involve only equations (3-1) and (3-11), with the final (after convergence) value of \mathbf{W} being used to form the L_p estimate of \mathbf{P} .

The sum of powers method (Equations (3-10) and (3-11)) has the advantage that knowledge of noise characteristics is used in the \mathbf{P} estimate (through the use of $\lambda_{\min} \mathbf{R}_b$); however, it has the disadvantage (compared to the enumeration method) that the analysis of Appendix A no longer applies (since an ensemble of vectors, rather than a single vector, is being estimated). On the other hand, experimental results using the sum of powers method indicate that it can be expected to converge at least as quickly as the enumeration method. For example, in the experiments described in Chapter IV, the average iteration count for the enumeration method was approximately 34, while the average iteration count for the sum of powers method was approximately 24, for a value of $p = 1$.

L_1 Selected Rows Method

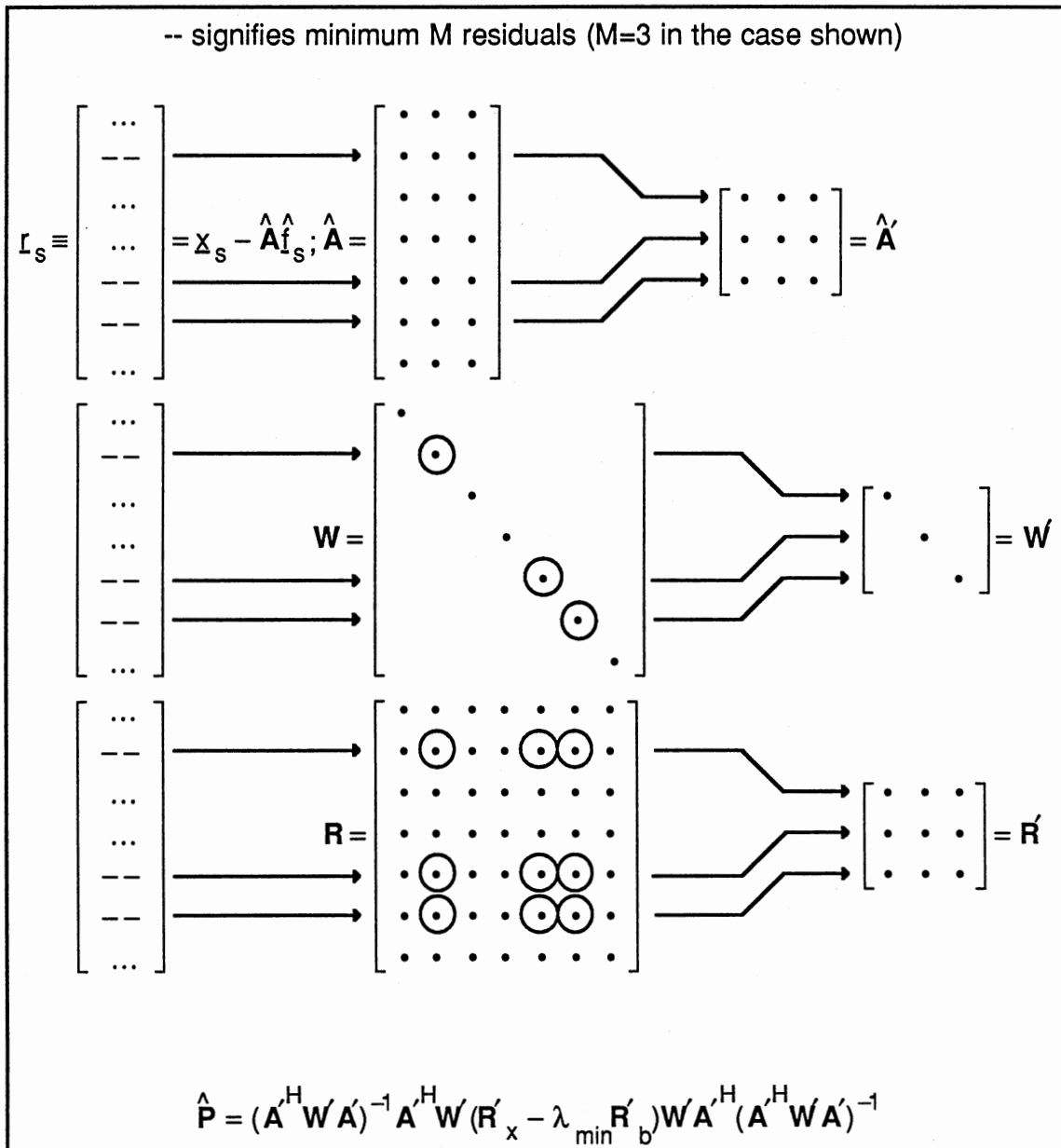
In a system of N equations with M unknowns, where $M < N$, L_1 solutions produced by the simplex algorithm in general satisfy M of the original equations

exactly. For the IRLS method, it is possible to produce a result which likewise satisfies M of the original equations exactly. Given this fact, it is reasonable to develop an approach which uses the exactly-satisfied entries in (1-6) to form an estimate of \mathbf{P} (for the L_1 case).

For example, the "sum of powers" approach described above produces a single N -element residuals vector (\mathbf{r}) as a result of the estimation process. Further, this vector is produced before an estimate of \mathbf{P} is computed. An estimate of \mathbf{P} , then, could be computed by solving the signal vector estimation problem for the M rows of \mathbf{A} which correspond to the smallest M entries in the final residuals vector. This approach will, in most cases, yield an exact L_1 solution to the selected M rows (Scales and Treitel, 1987). The desired estimate of \mathbf{P} would then be formed using the selected M rows of \mathbf{A} and \mathbf{W} , and the corresponding M^2 elements of \mathbf{R} , which correspond to the smallest M entries in the final residuals vector, as shown in Figure 2. A similar approach could be applied to the "enumeration" estimation method by selecting entries in \mathbf{A} , \mathbf{W} , and \mathbf{R} based, for instance, on the frequency (across the sample ensemble) with which a given set of M rows produces minimum residuals.

L_p Estimates of Directions of Arrival

As stated in Chapter II, MUSIC is a least squares, or L_2 , estimator of directions of arrival. In order to take full advantage of the L_p norm, it is desirable to develop an L_p direction of arrival estimator based on a combination of the principals reflected in MUSIC and those embodied by the IRLS algorithms above. This provides an L_p solution to the complete DOA problem, rather than just an L_p -based modification to the last phase (estimation of \mathbf{P}) of a solution based on L_2 principals (MUSIC).

Figure 2. L₁ Selected Rows Method

The form of an L_p DOA estimator is suggested by the expression for MU - SIC's DOA estimator (Equation (2-29)). The MUSIC algorithm estimates wave directions of arrival by locating the maxima of the function:

$f(\theta) = \{\underline{a}^H(\theta)\mathbf{E}\mathbf{E}^H\underline{a}(\theta)\}^{-1}$, or, equivalently, the minima of the function:

$$g(\theta) = \underline{a}^H(\theta)\mathbf{E}\mathbf{E}^H\underline{a}(\theta).$$

The expression for $g(\theta)$ can be rewritten in the following form:

$$\begin{aligned} g(\theta) &= \sum_{n=M+1}^N [\underline{a}^H(\theta)\underline{e}_n][\underline{e}_n^H\underline{a}(\theta)] \\ &= \sum_{n=M+1}^N [\underline{a}^H(\theta)\underline{e}_n][\underline{a}^H(\theta)\underline{e}_n]^* \\ &= \sum_{n=M+1}^N |\underline{a}^H(\theta)\underline{e}_n|^2 \end{aligned}$$

Clearly, MUSIC DOA estimates are just those angles which minimize a sum of squares of magnitudes of the dot products shown above. Since this is the classic form of an L_2 estimate (minimization of a sum of magnitudes squared), the proposed L_p DOA estimates minimize a sum of magnitudes raised to the p^{th} power, as follows:

$$g_p(\theta) = \sum_{n=M+1}^N |\underline{a}^H(\theta)\underline{e}_n|^p \quad (3-12)$$

Some observations about (3-12) are in order. First, the angles θ which minimize g_p are plainly L_p estimates of direction of arrival, in the sense that those angles minimize the p -norm of the difference between the array manifold ($\underline{a}(\theta)$) and the signal subspace ($\underline{e}_n, n = 1, 2, \dots, M$). Put another way, since subspace \underline{e}_n ($n = M+1, M+2, \dots, N$) forms a basis for the noise subspace, minimizing the sum of (3-12) minimizes (in the p -norm sense) the projection of the

vector $\underline{a}(\theta)$ onto the noise subspace. Further, since the noise subspace is the orthogonal complement of the signal subspace, the distance between the array manifold and the signal subspace is consequently minimized in the p -norm sense. That is, minimizing g_p selects the angles that best fit the observed data, in the p -norm sense, subject to the constraints of array geometry.

Second, no iteration is required to determine solutions of (3-12), as is necessary in the case of estimating \mathbf{P} . This implies that a \mathbf{P} estimate based on L_p DOA estimates (which might be considered a form of L_p estimate of \mathbf{P}) can be obtained with essentially no more computation than is required to obtain an L_2 estimate of \mathbf{P} from the MUSIC DOA estimates. Also, since no convergence issues are involved in minimizing g_p , one could obtain L_p DOA estimates (and the associated L_p \mathbf{P} estimate) for any value of $p > 0$ (although the interpretation for values of p outside the range $1 \leq p \leq 3$ might be open to question).

Third, since g_p provides a measure of the p -norm distance between the array manifold and the signal subspace, it gives at least a relative indication of the success of the estimation process. For example, if additional data is taken following an estimation step, and incorporation of the additional data results in lower values of g_p , it might reasonably be claimed that the estimates obtained using the additional data are in some sense "better" than the original estimates. Of course, such a qualitative discussion as that given requires considerable experimental or theoretical support if useful measures of goodness-of-fit are to be developed.

Finally, for the case of $N = M + 1$ (the number of arriving waves is one less than the number of antenna elements), the angles which minimize g_p are unchanged by varying p . This is because the sum in (3-12) consists of one term for this case, and, of course, the minimum of a single term occurs at the same point regardless of the exponent to which that term is raised (as long as the ex -

ponent is positive). This implies that, when estimation of the directions of arrival of one fewer waves than array elements is attempted (a condition which we will call "array saturation"), L_p DOA estimates are exactly the same as L_2 estimates, regardless of the (positive) value of p used. In this case the noise subspace consists of a line in N -space passing through the single noise eigenvector \underline{e}_N . Therefore, any local minimum of the projection of a vector (particularly the array manifold vector) onto the noise subspace (a line) can occur in only one direction, regardless of the norm used. When array saturation occurs, therefore, L_2 DOA estimates are the only ones of interest (although L_p estimates of \mathbf{P} may be of value). On the other hand, iterative DOA estimation (discussed later) may provide improved DOA resolution even in the presence of array saturation.

Noise Aspects

As mentioned in Chapter II, the L_2 norm is the generally accepted basis for estimation techniques, and it is the starting point for the development of L_p techniques. It is also the only approach necessary for linear systems in the presence of Gaussian noise. For the direction of arrival problem, however, the basic model, as previously stated, is not linear (although the model on which the eigenvector decomposition is based is, of course, linear).

In addition to the nonlinearity aspect of the problem, it is not reasonable to suppose that all noise to which an antenna array is subjected will necessarily be Gaussian (indeed, as discussed in Chapter II, it may be unreasonable to assume that Gaussian noise is present in any estimation environment). As an example, consider the case of an array in the presence of a jamming signal. It is highly unlikely that the jammer's modulation envelope resembles Gaussian noise, and yet it might be desirable to treat the jammer as a noise source rather than as an additional signal (to facilitate analysis of the other signals of interest,

for instance). If the jammer is in fact emitting periodic pulses of short duration and high intensity, the signal received by the array could resemble impulsive noise, rather than Gaussian noise.

Another example of impulsive noise would be bursts of static either received by the array or generated within the receivers themselves. Other noise distributions are also possible; for example, continuous static might give rise to a uniform noise distribution. Also, even if the underlying noise sources are Gaussian, the samples might not reflect this fact (due, for example, to small sample size).

Since non-Gaussian noise is a distinct possibility in the DOA determination problem, the use of other than the L_2 norm seems a potential area of benefit. For example, the L_1 norm gives generally superior (to L_2) results in the presence of impulsive noise. Schroeder (1985) demonstrated that use of the L_p norm, $1 \leq p < 2$ permitted resolution of more closely spaced spectral components of a signal in non-Gaussian noise than did the L_2 norm. Figure 3 provides some illustrations of the resolution improvement (over L_2 -norm performance) possible when the L_1 norm is used rather than the L_2 norm in the linear prediction approach to spectral estimation used by Schroeder.

The research described herein considered Gaussian, uniform, and impulsive noise distributions. Results of DOA estimation using the L_1 norm in the presence of impulsive noise were in general superior to results using the L_2 norm, as expected (Chapter IV).

Adaptive Methods

As mentioned above, the value of g_p yields some information concerning the degree of success of the DOA estimation process. Other measures of algorithm performance can also be computed. For example, Schmidt points out that

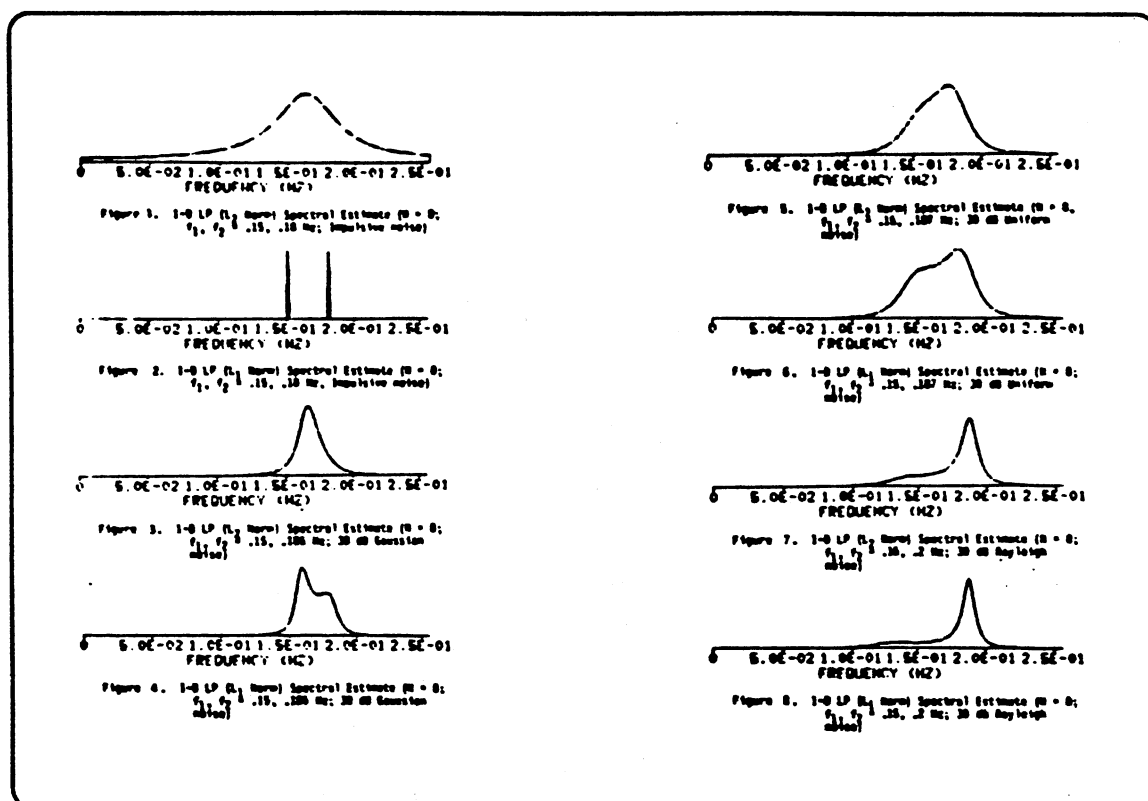


Figure 3. Resolution of Spectral Components in Noise (Schroeder, 1985)

it is possible for the MUSIC algorithm to indicate the presence of more arriving waves than can be resolved due to wave proximity. That is, if two waves are very closely spaced, it is possible for the MUSIC algorithm to be able to resolve the direction of arrival of only one of them, but still provide an indication that more are present.

Two estimates of the number of waves arriving at an antenna array are given in Equations (3-13) and (3-14). Schmidt (1981) suggests a χ^2 -based likelihood ratio test for the number of arriving waves, as shown in Equation (3-13):

$$\text{LRT}(M) = S(N - M) \ln \frac{\lambda_a}{\lambda_g} \approx \chi_d^2 \quad (3 - 13)$$

$$d \equiv \frac{1}{2} [(N - M) - 1][(N - M) + 2]$$

where:

S = number of antenna array output samples taken.

N = number of antenna array elements.

M = number of arriving waves hypothesized.

λ_a = arithmetic mean of the smallest N - M generalized eigenvalues.

λ_g = geometric mean of the smallest N - M generalized eigenvalues.

An alternate estimate, which provides better results in some situations, is advanced by Marple (1987). In this approach, the number of arriving waves is estimated as the value which produces a minimum in the Akaike Information Criterion (AIC) as modified by Wax and Kailath:

$$\text{AIC}(M) = (N - M) \ln \left(\frac{\frac{1}{N - M} \sum_{m=M+1}^N \lambda_m}{\prod_{m=M+1}^N \lambda_m^{-(N-M)}} \right) + M(2N - M) \quad (3 - 14)$$

where N and M are as defined for Equation (3-13), and λ_m is the m^{th} generalized eigenvalue (λ_1 being the largest and λ_N the smallest).

The above expressions provide estimates only - there is no guarantee that either expression will produce an accurate incoming-wave count in a given situation. The question of how many waves are being received at an antenna array (or how many spectral components are present in a noisy medium) is one of continuing interest. However, since the research described herein addresses the direction of arrival estimation process itself, the number of arriving waves is assumed to be known or estimated in advance.

Given a situation in which more impinging waves can be detected than can be located, it is important to investigate approaches which make use of performance measures of some sort to alter the behavior of the overall estimation process. Lansford (1988) investigated the use of kurtosis as a performance measure in the application of linear predictive models to the speech recognition problem. The nature of the measures used, as well as the specific modifications to be made to the algorithms, of necessity depend heavily on the application to be supported. Since each application involves its own engineering trade-offs, a modification that would constitute an improvement in performance in one application might actually be a degradation in another.

As an example, consider the jammer problem mentioned above. In one application, the DOA and signal characteristics of the jammer might be of interest, in which case the jammer should be considered an additional signal source. On the other hand, in a different application it might be desirable to treat the jammer as a noise source, in which case determination of its DOA and signal characteristics is unnecessary and might even interfere with problem solution.

Although any adaptive method is subject to application constraints, some can be discussed which would likely find use in a number of different scenarios. Such techniques as iterative revision of eigenvectors or DOA estimates fall into this category.

Iterative Direction of Arrival Revision

The DOA spectrum produced using (2-29) typically exhibits a number of peaks corresponding to estimated directions of arrival. In the case of multiple closely spaced arriving waves, fewer peaks than arriving waves may appear.

An example is shown in Chapter IV wherein two arriving waves at $\pm 30^\circ$ effectively suppress a third peak which should appear at 0° .

In such situations, an approach whereby the effects of previously located waves could be removed from the DOA spectrum might permit resolution of additional spectral peaks. Such a "spectral subtraction" can be accomplished by forming estimates of the array phase shift matrix (\mathbf{A}) and the source signal covariance matrix (\mathbf{P}) for the wave(s) to be removed from the spectrum, subtracting the product $\mathbf{A}\mathbf{P}\mathbf{A}^H$ from the original array output covariance matrix (\mathbf{R}_x), and forming a new eigenvalue problem from the remainder.

Unfortunately, the difference mentioned above is almost never positive semidefinite, which of course implies that it is not a valid covariance matrix. This can be explained by noting that, since only a subset of the arriving waves has been identified, the diagonal elements of the \mathbf{P} estimate (which are estimates of arriving wave power) are very likely larger than the corresponding actual elements in \mathbf{P} . Thus, the term $\mathbf{A}'\mathbf{P}'\mathbf{A}'^H$, where \mathbf{A}' and \mathbf{P}' are the estimates of \mathbf{A} and \mathbf{P} , respectively, tends to "overcompensate" for the waves for which the estimates were made, producing negative eigenvalues in the difference.

In order to eliminate the undesirable effect of producing negative eigenvalues, some real fraction of the product estimate can be subtracted from the observation covariance matrix, as follows:

$$\mathbf{R}_x' = \mathbf{R}_x - \gamma\mathbf{A}'\mathbf{P}'\mathbf{A}'^H; \quad 0 < \gamma \leq 1 \quad (3 - 15)$$

Naturally, the question arises, "What value should be used for γ ?" Since positive semidefiniteness is required of any covariance matrix (including \mathbf{R}_x'), the optimum γ is the maximum value for which the difference (3-15) remains positive semidefinite. In order to determine this optimum value in a theoretical sense, it is necessary to express all principal minors of the difference (3-15) in

terms of γ . That is, the formation and solution of a polynomial in γ of at least N^{th} order (where N is the number of antenna elements), for an arbitrary matrix pair $\mathbf{R}_x, \mathbf{A}'\mathbf{P}'\mathbf{A}'^H$, "suffices" to determine the optimum value of γ . Of course, this involves *at least* as much computation as did the original DOA estimation process itself!

This being the case, an alternative approach to determination of the optimum γ value is proposed. Although entirely empirical in nature, the alternative approach gives the optimum γ value to whatever precision is desired. The method implements a numerical binary search over the interval $(0,1]$ for the largest acceptable value of γ . That is, a value of 1 is tried, followed by 0.5, followed by either 0.25 or 0.75, and so on until the desired resolution is achieved. The value of the next "guess" at each stage is determined based on whether positive semidefiniteness was obtained at the previous stage. This approach requires a positive semidefiniteness test at each step. Although this can represent a considerable computational burden, the polynomial solution effort is avoided, which in itself is expected to reduce overall computational load. Additionally, the procedure can be stopped at any point, providing for an adjustable processing time (at the expense of low γ resolution for short processing times). In practice, relatively little resolution for γ may be required, since DOA resolution can be expected to improve to some extent with any reduction in dominant spectrum peaks. The extent to which such peak reduction can be expected to help depends strongly, of course, on the actual data involved. Several examples are provided in Chapter IV of the iterative DOA revision techniques discussed above.

A second concern in implementing the spectral subtraction described above is the danger of introducing spurious peaks. As an illustration of how this is possible, consider the operation of a band-pass filter with white noise applied

to its input. Since white noise contains energy at all frequencies, the filter will produce an output signal at (about) its center frequency, even though no signal was present at its input. A similar process is possible in digital signal processing, where false periodicities can be established using successive smoothing and differencing operations. Yarlagadda and Hershey (1987) point out that "... the creation of a false periodicity in this manner is related to the Slutsky-Yule effect that predicts the period of the false oscillation based upon the smoothing and differencing operations."

Other Adaptive Approaches

Another iterative procedure suggested by the MUSIC eigenvalue problem (Equation 2-27) involves the estimation of λ_{\min} and \mathbf{R}_b . The usual estimates of these parameters are formed from noise samples (assuming they are available) as follows:

$$\mathbf{R}_w = \sum_{s=1}^S \mathbf{w}_s \mathbf{w}_s^H \quad (3-16)$$

$$\lambda_{\min} = \text{tr}\{\mathbf{R}_w\} / N$$

$$\mathbf{R}_b = \mathbf{R}_w / \lambda_{\min}$$

Obviously, the accuracy of the above values of λ_{\min} and \mathbf{R}_b depends on the number of noise samples available and how representative of the underlying random process the available noise samples are. The effect of inaccuracies in the estimates (3-16) is to produce an \mathbf{R}_b value which, when used in (2-27), yields a value for λ_{\min} different from the above estimate. An adaptive approach which might improve the accuracy of the eventual eigenvalue problem solution is to compute the average of the λ_{\min} estimate from (3-16) and the actual λ_{\min} from (2-27), use the average to compute a new \mathbf{R}_b estimate (using (3-16)),

solve (2-27) again, and so on until the λ_{\min} estimate from (3-16) is acceptably close to the estimate from (2-27). This approach might be particularly useful in the case of small sample size, where an \mathbf{R}_b estimate is desired with relatively few \underline{w}_s samples.

In another adaptive procedure, iterative revision of the eigenvectors for DOA estimation is proposed (Fuhrmann, 1987). This approach employs an iterative singular value decomposition algorithm to periodically update estimates of the signal subspace vectors, and is most directly applicable to the case of time-varying directions of arrival. The subspace vector update is interleaved with both acquisition of new observation vector samples and DOA spectrum evaluation, the interleave ratios being determined by the application. The iterative DOA revision technique described above could be included directly into this procedure, allowing the decision concerning the number of waves to identify to be made dynamically.

CHAPTER IV

EXPERIMENTAL RESULTS

Several procedures described in previous chapters were implemented on a general purpose computer, and tested using simulated data sets. The approaches tested were the L_p DOA estimation algorithm, the iterative DOA revision approach, and the enumeration and sum of powers \mathbf{P} estimation procedures.

To solve the generalized eigenvalue problem (2-27), a generalized singular value decomposition algorithm due to Van Loan was employed (Appendix B) (Van Loan, 1985). Further, due to the similarity of the form of the matrix term in (3-1) to the Moore-Penrose pseudo-inverse, a modified Moore-Penrose pseudo-inverse algorithm (Appendix C) was employed to compute L_p estimates of source signal vector samples (\mathbf{f}_s) (Graybill, 1969).

The program described in Appendix D was used to generate simulation data for arrays of arbitrary geometry illuminated by arbitrary wave patterns in the presence of either correlated or uncorrelated noise. Noise generators with fixed, uniform, Gaussian, and impulsive distributions were supported.

Effects of Noise Power on DOA Resolution

Figures 4 through 8 illustrate the effects of impulsive noise on the DOA spectra produced by the L_p MUSIC algorithm. Figure 4 shows the antenna array and impinging wave pattern simulated. Figures 5 through 8 show that, as

SNR is dropped progressively from 20db to 0db, DOA resolution fails for decreasing values of p , until (Figure 8) $p = 1$ is the only value for which resolution is obtained. The impulsive noise was applied at an average of every 20 samples in the experiments shown in Figures 5 through 7, and 25 samples in Figure 8.

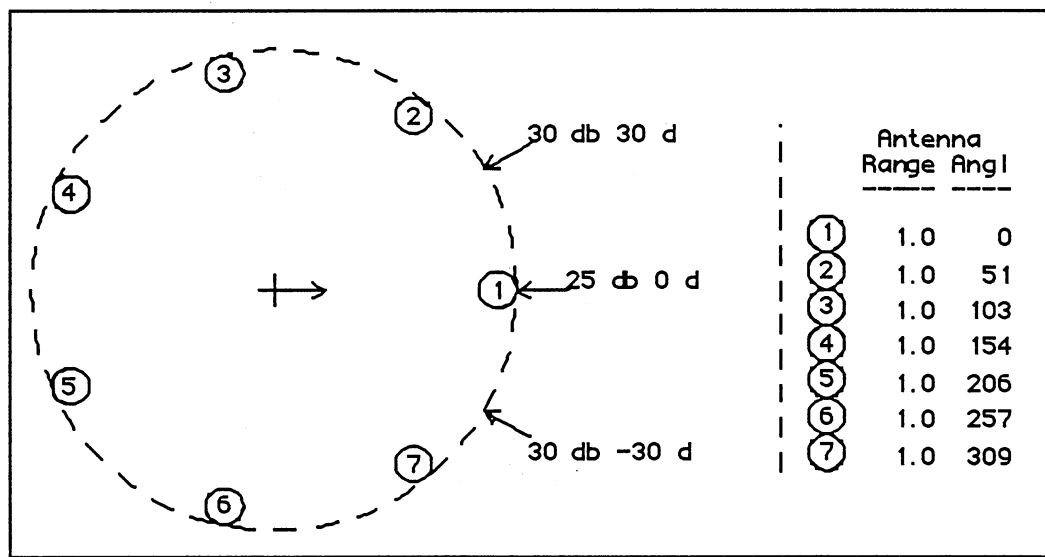


Figure 4. Antenna Array Geometry, Configuration #1

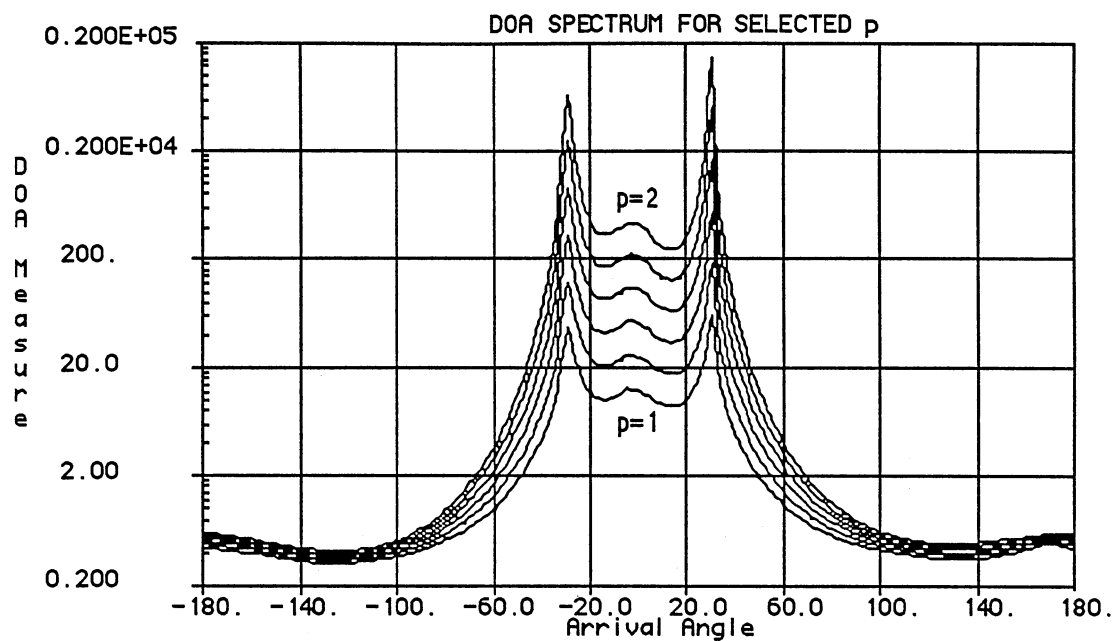


Figure 5. DOA Spectra, SNR = 20db, 20-Sample Interval

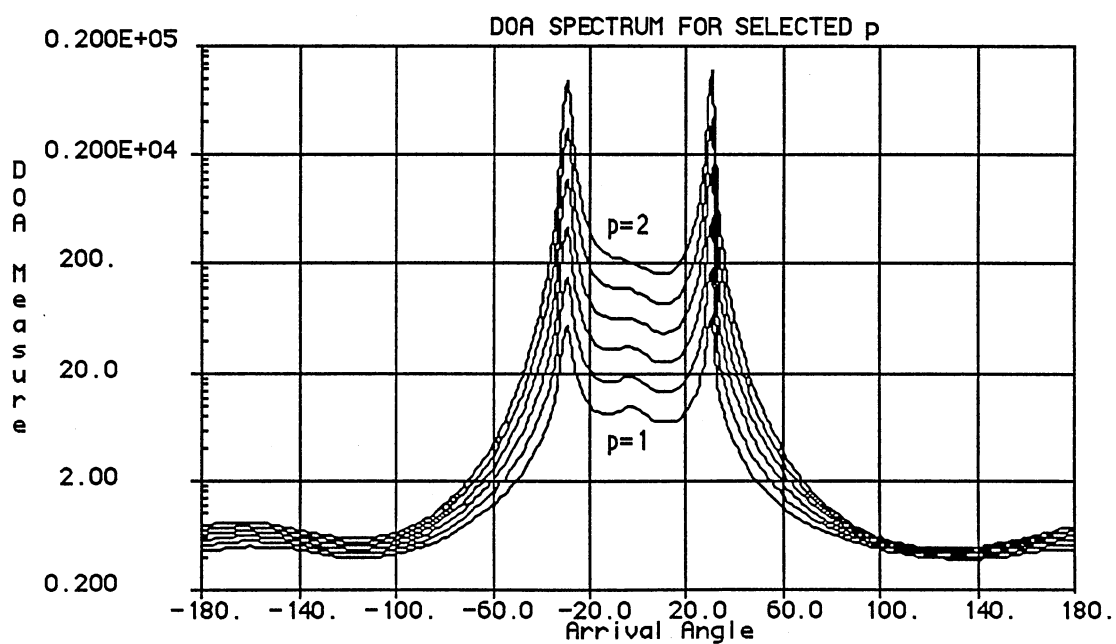


Figure 6. DOA Spectra, SNR = 15db, 20-Sample Interval

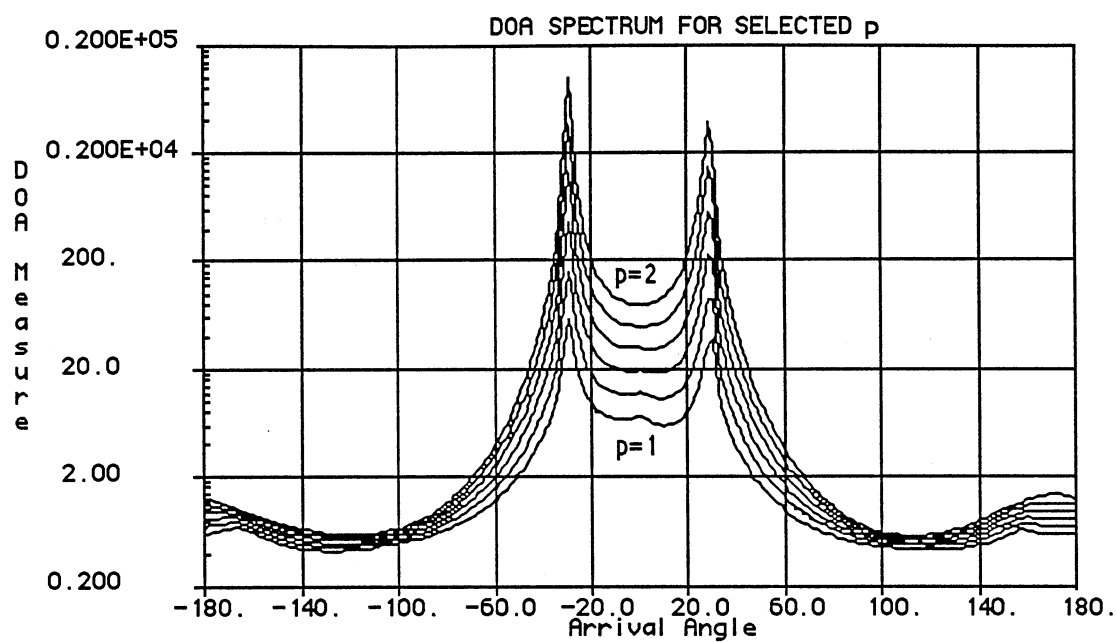


Figure 7. DOA Spectra, SNR = 10db, 20-Sample Interval

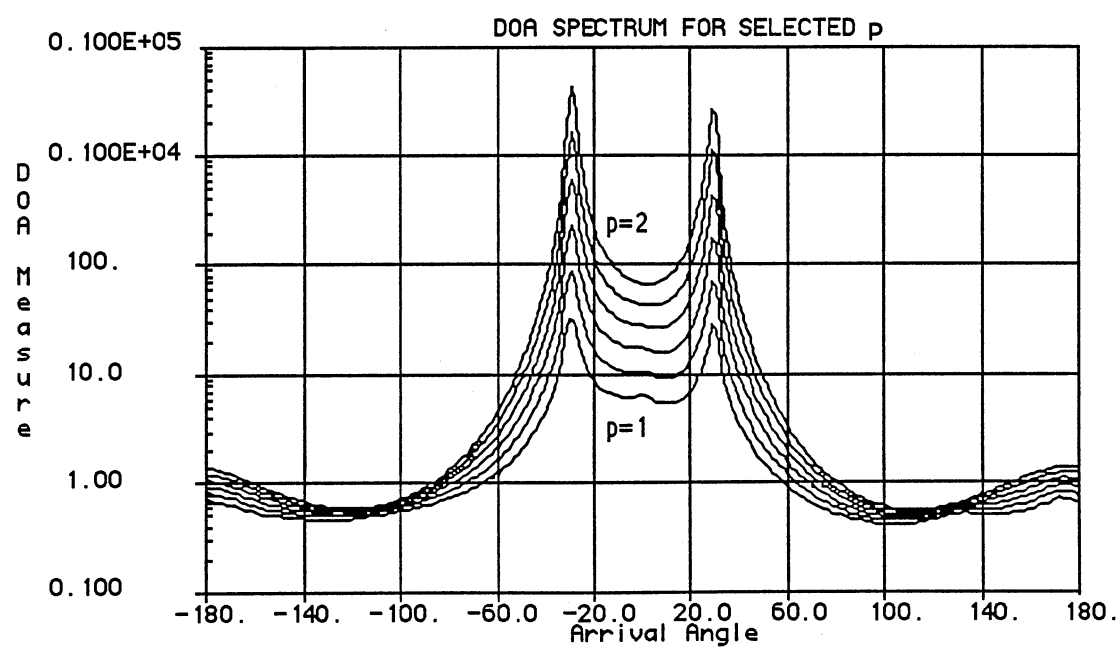


Figure 8. DOA Spectra, SNR = 0db, 25-Sample Interval

Figures 9 and 10 illustrate DOA spectra for values of p between 2 and 3. Noise strengths (relative to maximum incoming signal strength) and average impulse intervals were -20db each 20 samples for Figure 9 and 0db each 20 samples for Figure 10. Arrays as illustrated in Figure 4 were simulated in both cases.

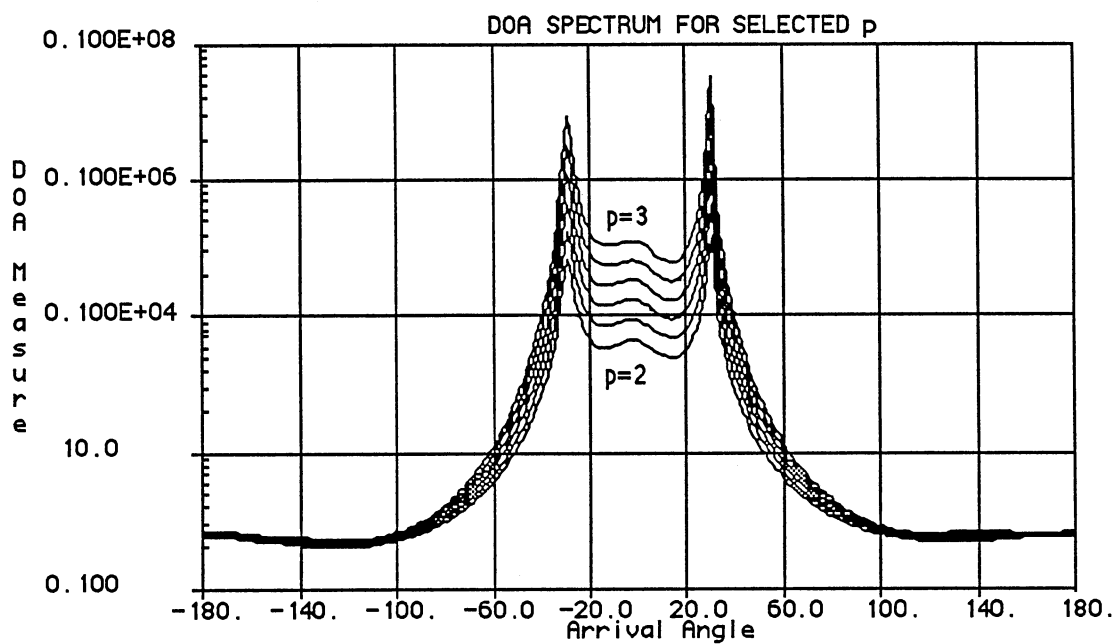


Figure 9. DOA Spectra, SNR = 20db, 20-Sample Interval

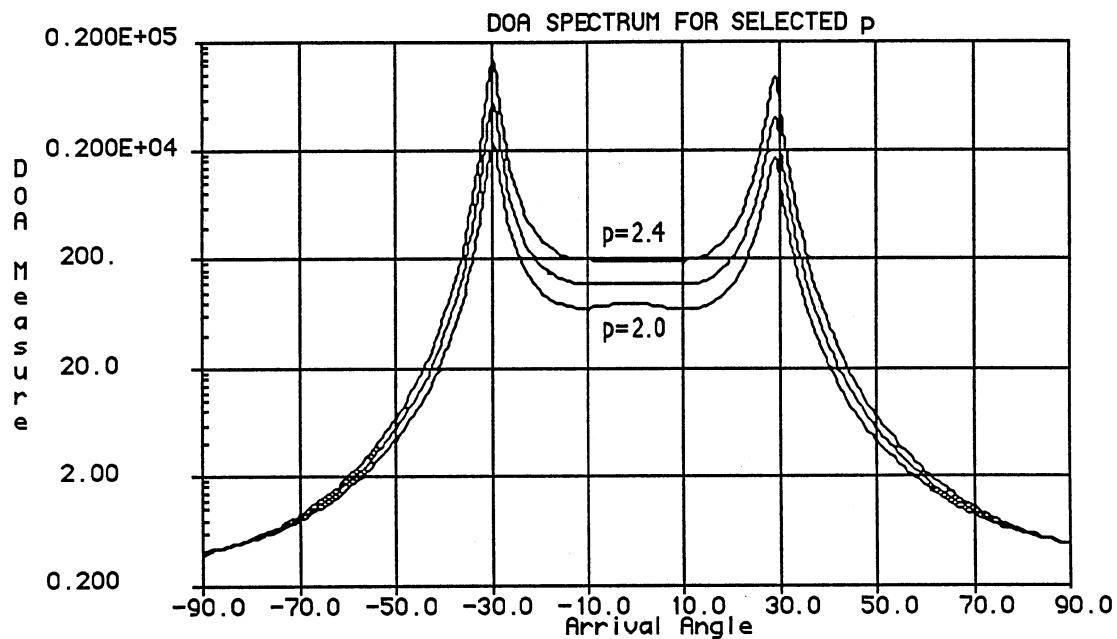


Figure 10. DOA Spectra, SNR = 0db, 20-Sample Interval

Figure 11 shows a second simulated array/wave combination, and Figures 12 through 14 illustrate a similar trend of progressive DOA resolution failure with decreasing SNR.

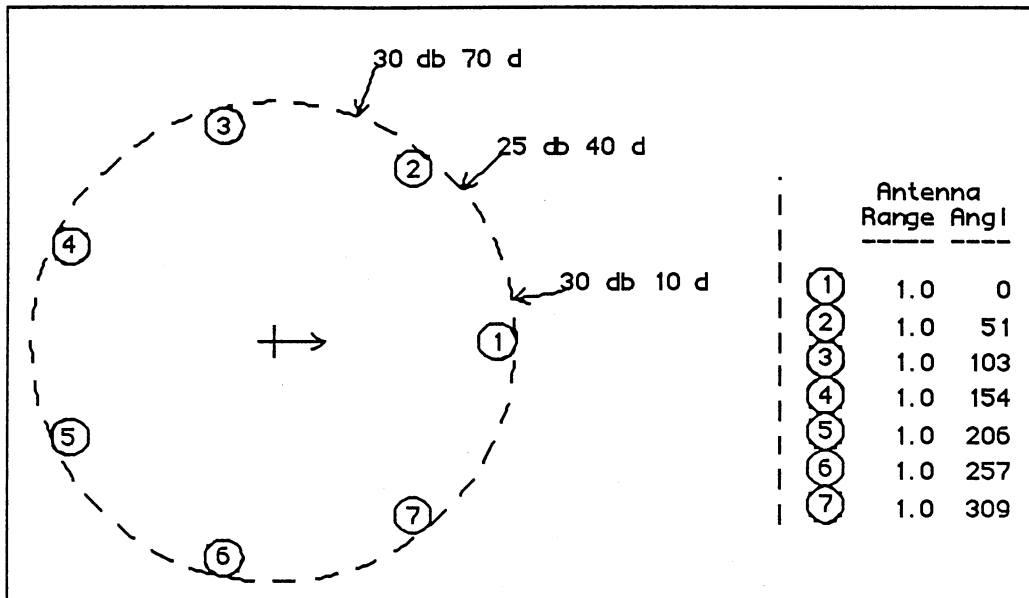


Figure 11. Antenna Array Geometry, Configuration #2

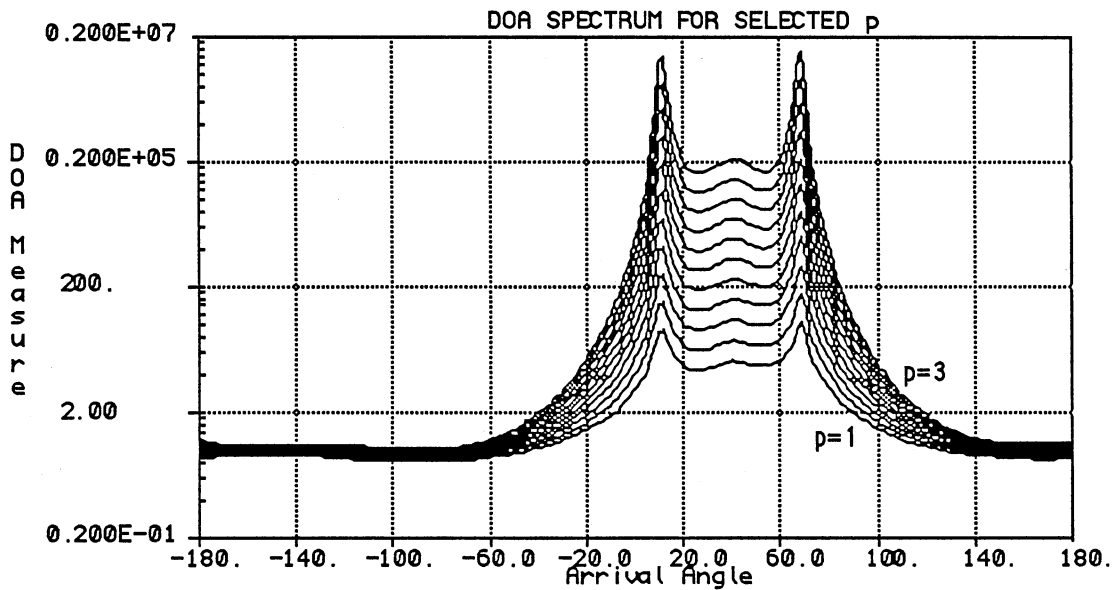


Figure 12. DOA Spectra, SNR = 30db, 50-Sample Interval

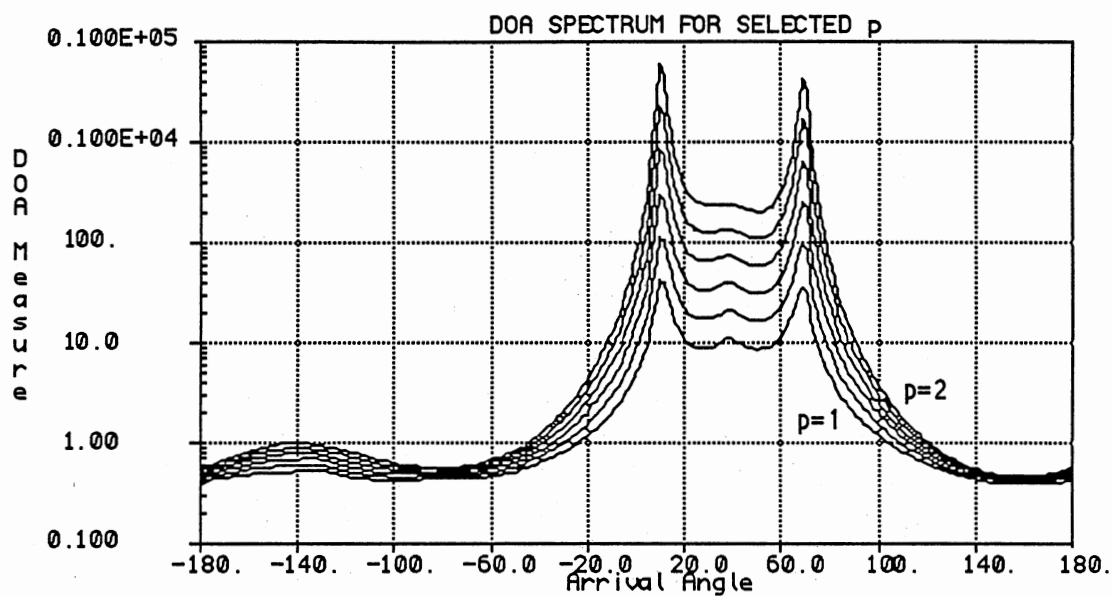


Figure 13. DOA Spectra, SNR = 20db, 50-Sample Interval

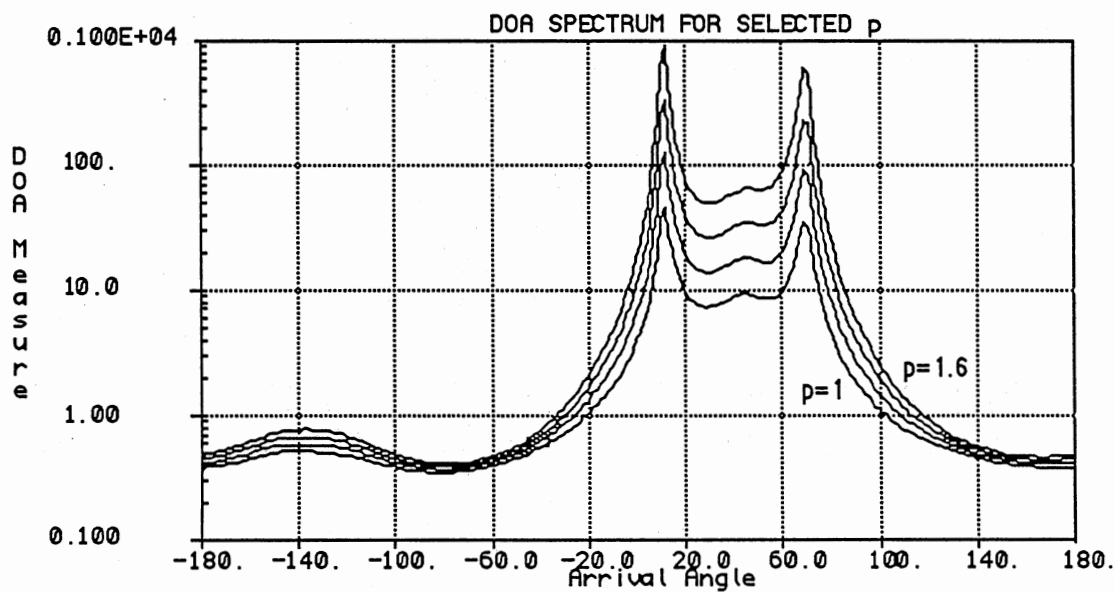


Figure 14. DOA Spectra, SNR = 10db, 50-Sample Interval

Effects of Noise Interval on DOA Resolution

Figures 15 through 18 illustrate that, as the average interval between impulsive noise bursts is increased, DOA resolution shows a general improvement. The arrays simulated were as illustrated in Figure 11.

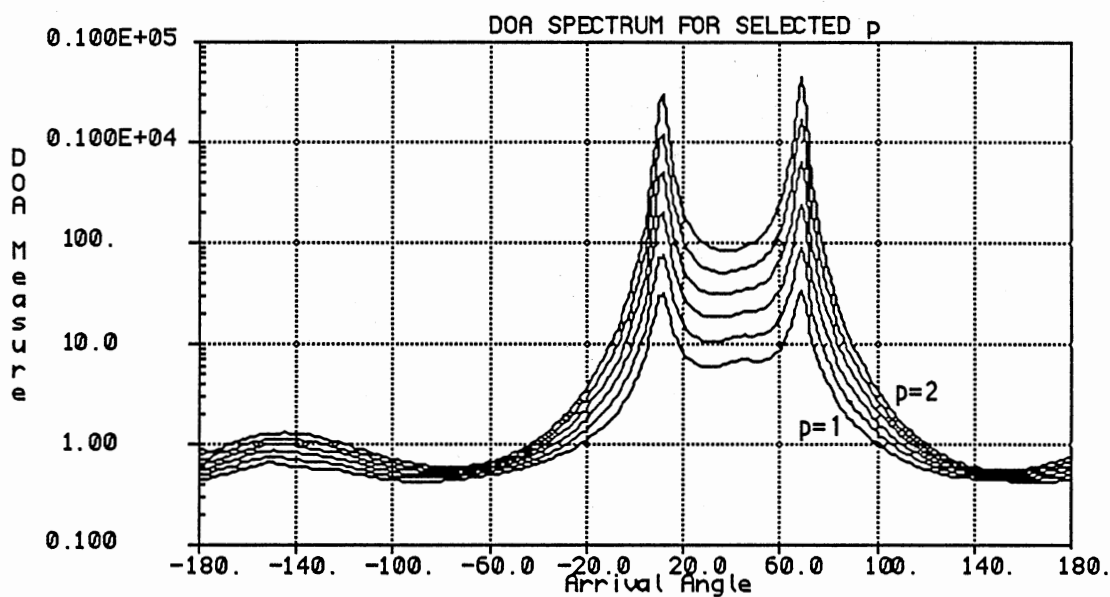


Figure 15. DOA Spectra, SNR = 20db, 40-Sample Interval

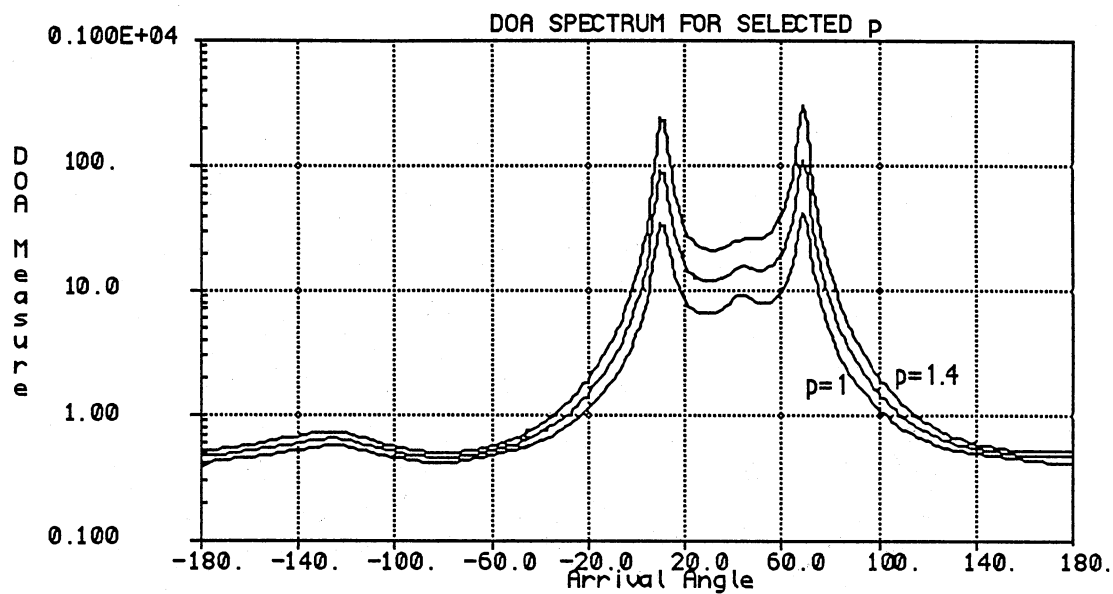


Figure 16. DOA Spectra, SNR = 20db, 45-Sample Interval

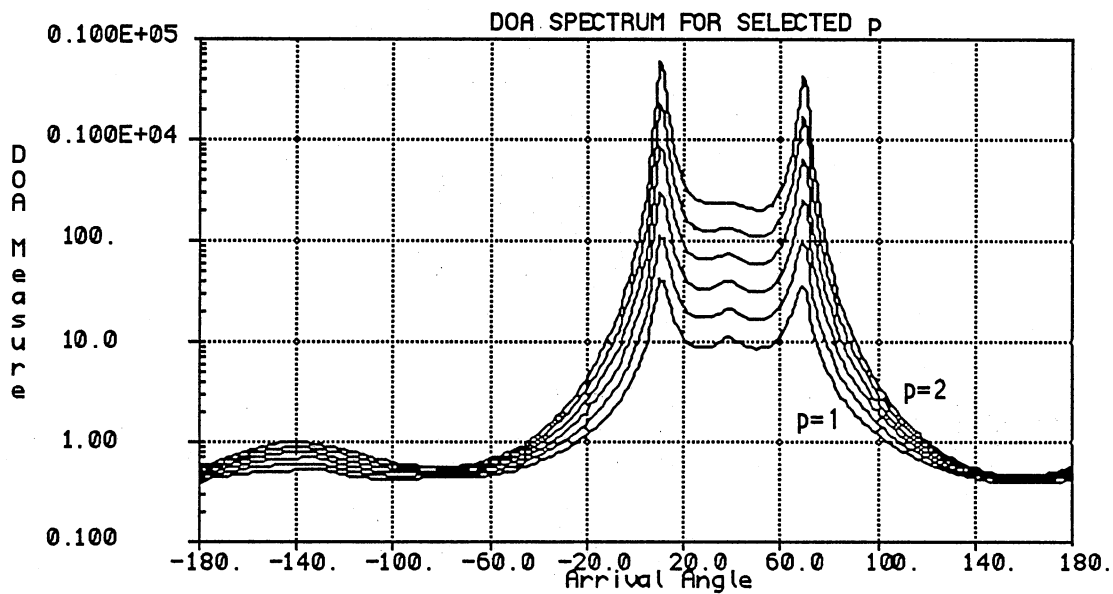


Figure 17. DOA Spectra, SNR = 20db, 50-Sample Interval

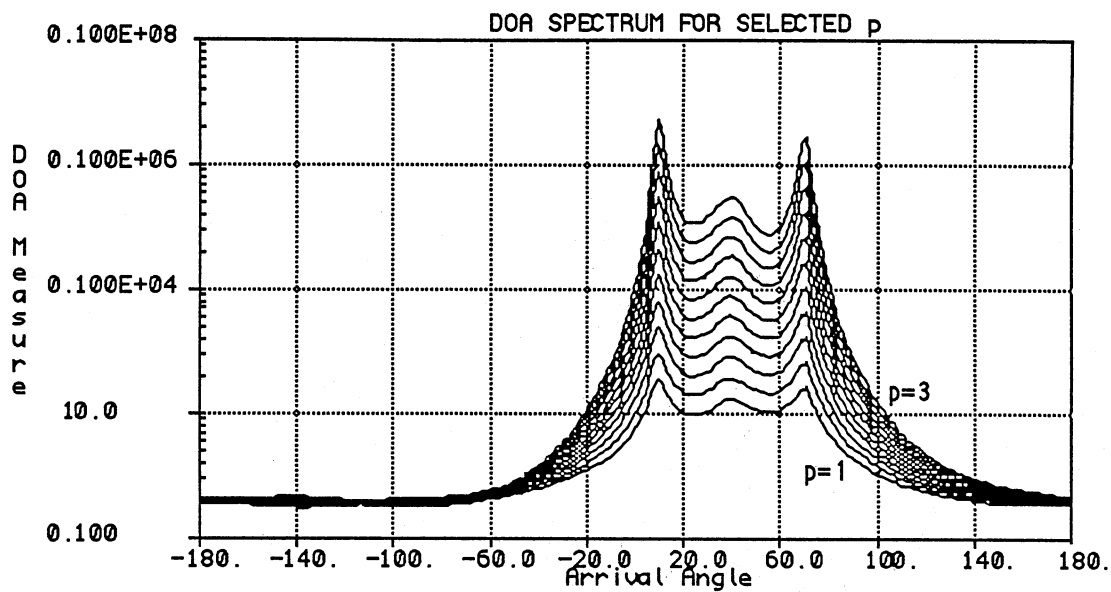


Figure 18. DOA Spectra, SNR = 20db, 55-Sample Interval

Iterative DOA Revision

Figures 19 - 22 provide an illustration of the application of the iterative DOA revision techniques discussed in Chapter III. Figure 19 shows the antenna array/arriving wave geometry simulated. The array was subjected to simulated impulsive noise at -20db (relative to maximum wave arrival strength) and a 40-sample average interval. As shown in Figure 20, resolution of the wave arriving from 35° is achieved only for values of $p < 2.0$. Figure 21 shows the DOA spectra which result from spectral subtraction of the effects of the wave arriving from 0° .

Figure 21 was produced as follows. The spectrum for $p = 2$ in Figure 20 (which does not resolve the wave from 35°) was used to estimate the angle of arrival of the wave from 0° . This DOA estimate was used to generate estimates of the **A** and **P** matrices corresponding to the identified arriving wave (from 0°).

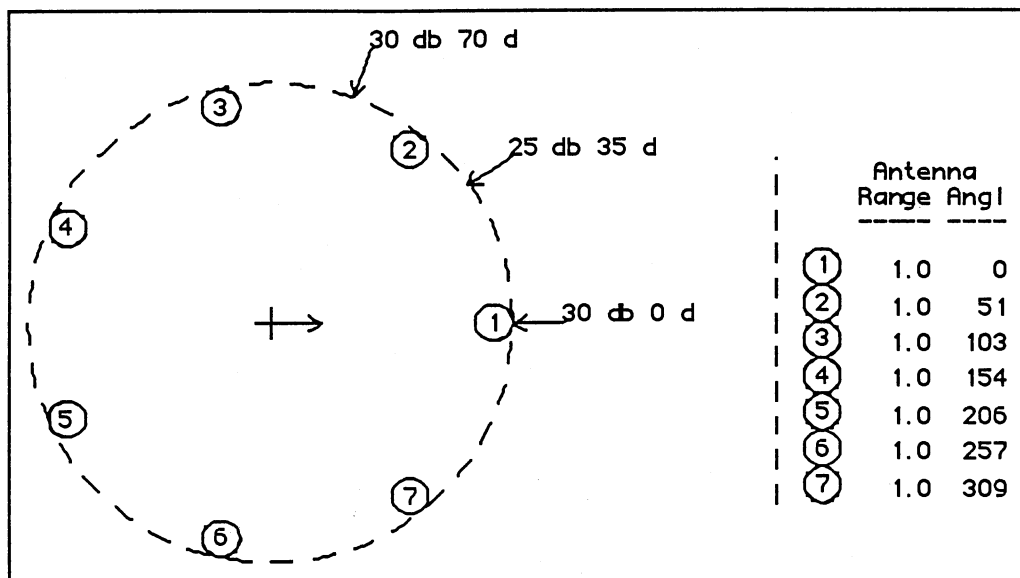


Figure 19. Antenna Array Geometry, Configuration #3

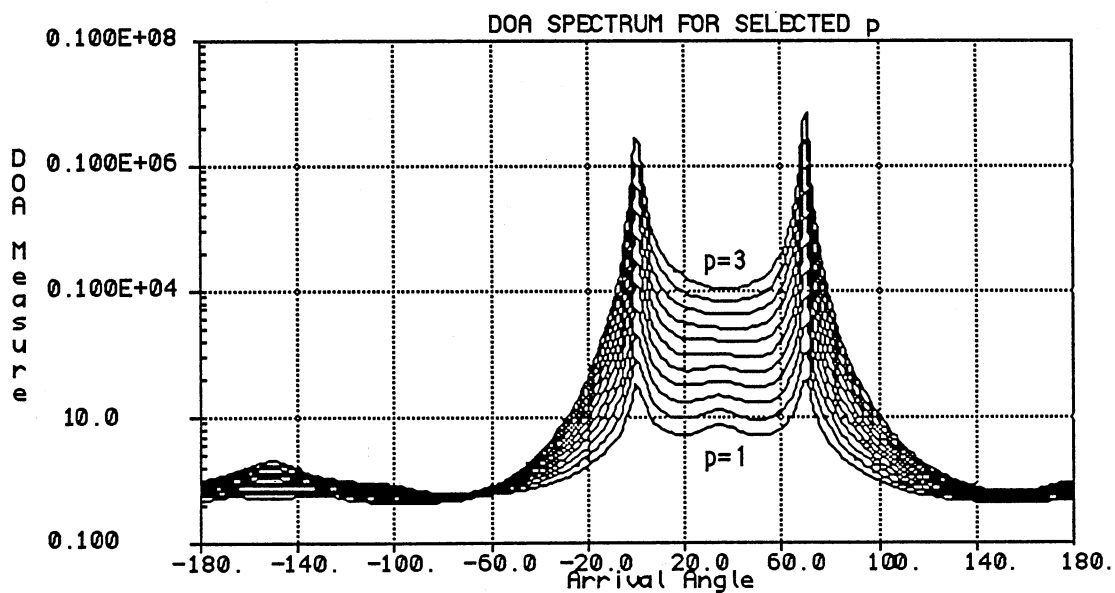


Figure 20. DOA Spectra, SNR = 20db, 40-Sample Interval

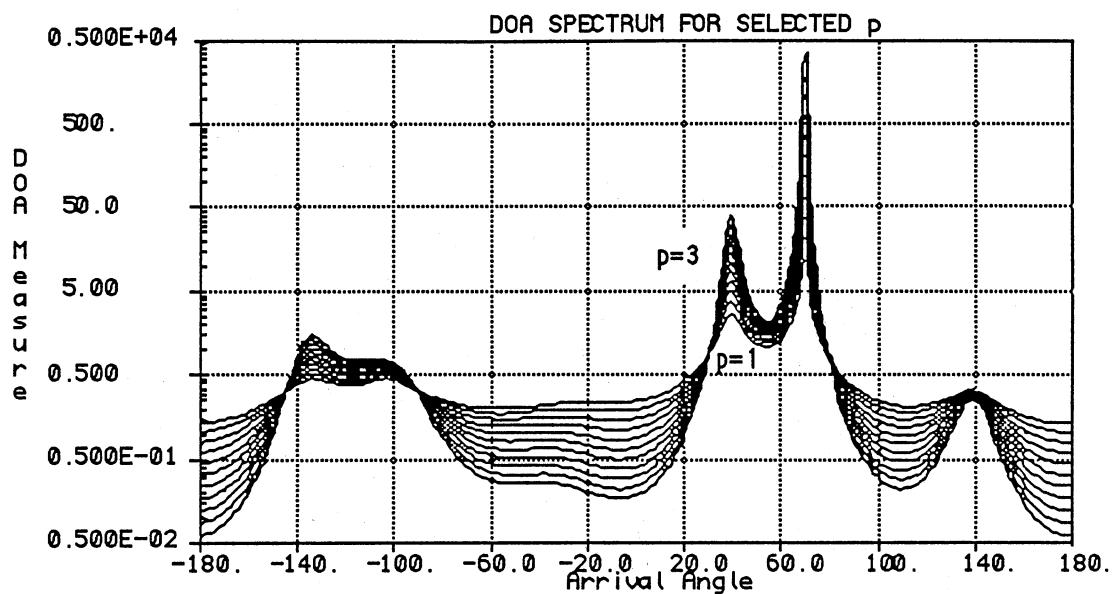


Figure 21. Configuration #3 Spectra Following Covariance Revision

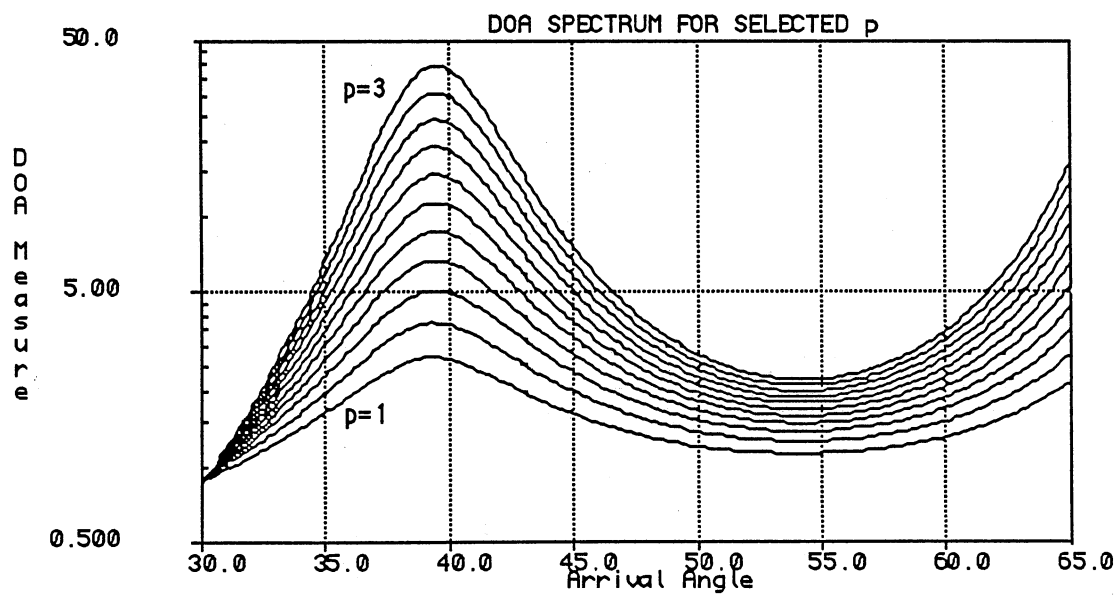


Figure 22. Configuration #3 Revised Spectra, 30° to 65°

These estimates were in turn used to revise the observation covariance matrix (\mathbf{R}_x), and a new eigenvalue problem involving the revised \mathbf{R}_x was solved to produce the spectra in Figure 21. Figure 22 is a more detailed view of the region in Figure 21 from 30° to 65° .

Figures 21 and 22 illustrate both the benefits and some of the potential hazards of the iterative DOA revision technique. As can be seen from the figures, the previously unresolved wave from 35° is resolvable for $1 \leq p \leq 3$ following covariance revision. On the other hand, since covariance revision emphasizes all peaks in the DOA spectrum (except those being used in the revision process), the (less prominent) peaks at $+140^\circ$ and between -90° and -150° , which do not correspond to actual wave arrivals, are emphasized as well. So, as mentioned in Chapter III, when applying iterative DOA revision one must use caution to avoid the introduction of spurious wave arrival indications.

In producing the spectra shown in Figures 21 and 22, the optimum multiplier for the \mathbf{APA}^H estimate (γ in Equation (3-15)) was resolved to 8 binary digits (bits). This implies that the value used was within 1% of the value which would have been available using infinite precision. As discussed in Chapter III, the use of a γ estimate with lower resolution should produce some DOA resolution improvement, while saving computation time. Figures 23 through 25 illustrate the effects on DOA spectra of using a γ estimate which is resolved to only 2 bits (within 50% of the value available using infinite precision).

Figure 23 shows the antenna array/arriving wave geometry simulated. The array was subjected to simulated impulsive noise at 0db (relative to maximum wave arrival strength) and a 35-sample (average) interval. Figure 24 shows that the two waves arriving from 35° and 70° produce only one spectrum peak, at about 65° . Figure 25 shows the result of revising the observation covariance matrix using estimated DOAs (from the $p = 2$ spectrum) for the waves

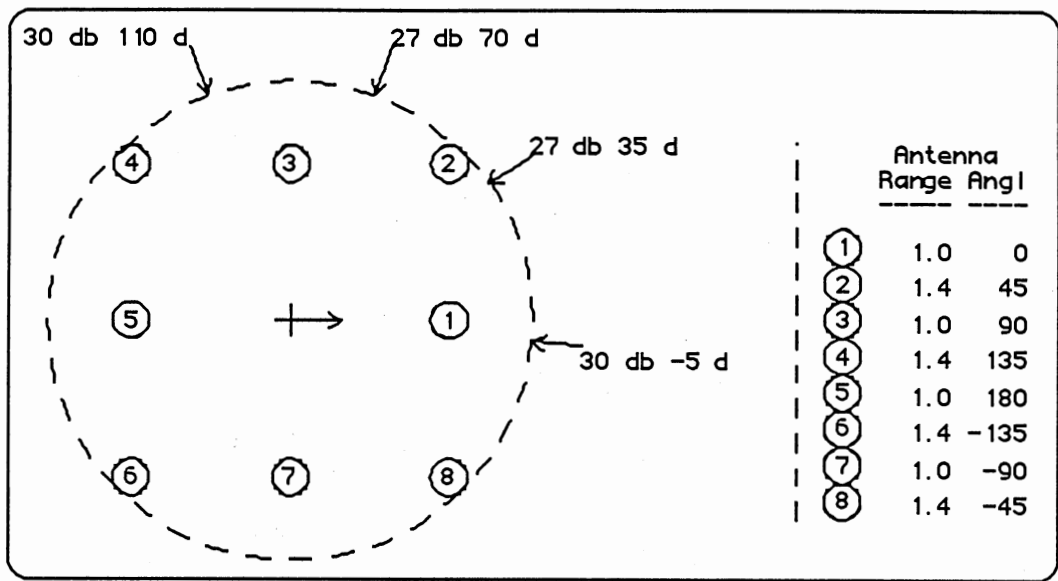


Figure 23. Antenna Array Geometry, Configuration #4

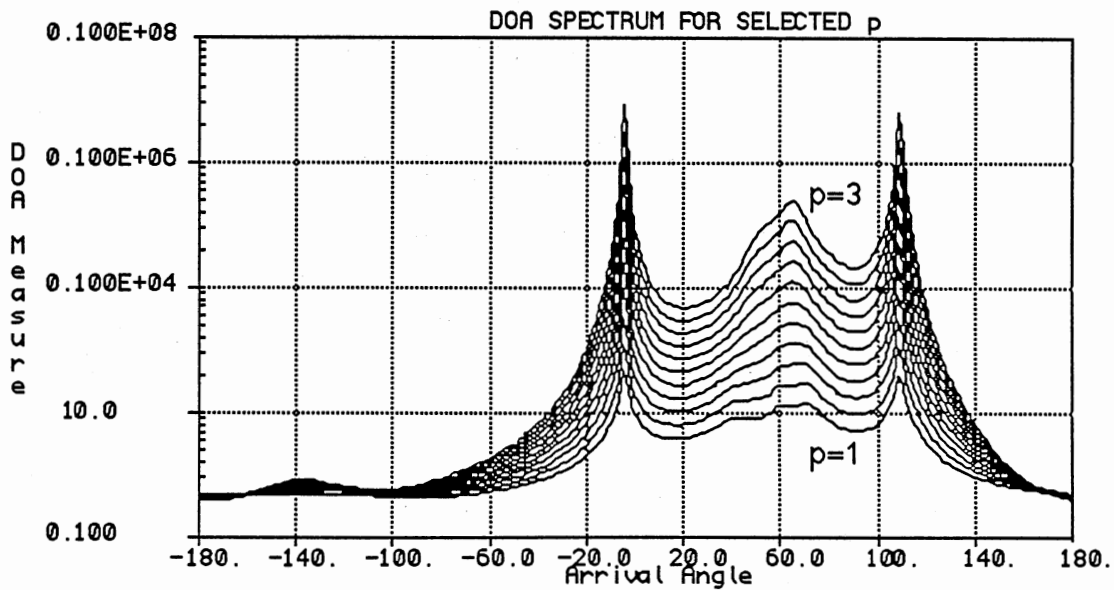


Figure 24. DOA Spectra, Configuration #4

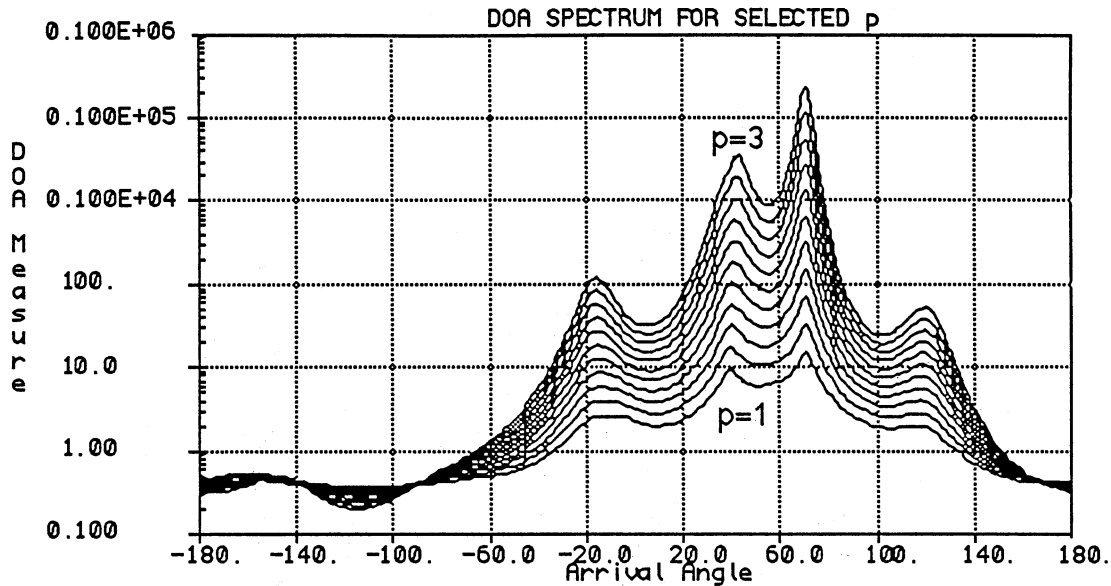


Figure 25. Configuration #4 Spectra Following Covariance Revision

arriving from -5° and $+110^\circ$. The revision was made using a γ estimate which was correct to only 2 bits. As shown in Figure 25, DOA resolution for both interior waves is achieved following covariance revision for values of p over the entire range from 1 to 3.

The execution time required to obtain the value of γ used to produce Figures 21 and 22 (eight bit precision) was 1.3 CPU seconds on a VAX 11/750 processor. The eight-bit-precision value obtained was 0.82421875. To obtain a value precise to two bits ($\gamma = 0.75$) for the same problem required 0.92 CPU seconds, a 29% time savings. On the other hand, the time for computation of the \mathbf{P} matrix estimate was 0.98 CPU seconds for $p = 2$, and 25.43 CPU seconds for $p = 1$. Thus, the time required to obtain even a comparatively high resolution estimate of γ can be small compared to the time required to estimate \mathbf{P} . For the experiment shown in Figures 23 through 25, estimation of the \mathbf{P} matrix ($p = 2$)

required 1.71 CPU seconds, and determining γ (to two bits) required 1.22 CPU seconds.

Figure 26 shows values of γ as a function of the ratio between the maximum diagonal element of the \mathbf{P} estimate and the maximum diagonal element of \mathbf{R}_x . It is clear from the figure that, although there might be a tendency for γ to drop as the mentioned ratio increases, the actual functional relationship is extremely complicated.

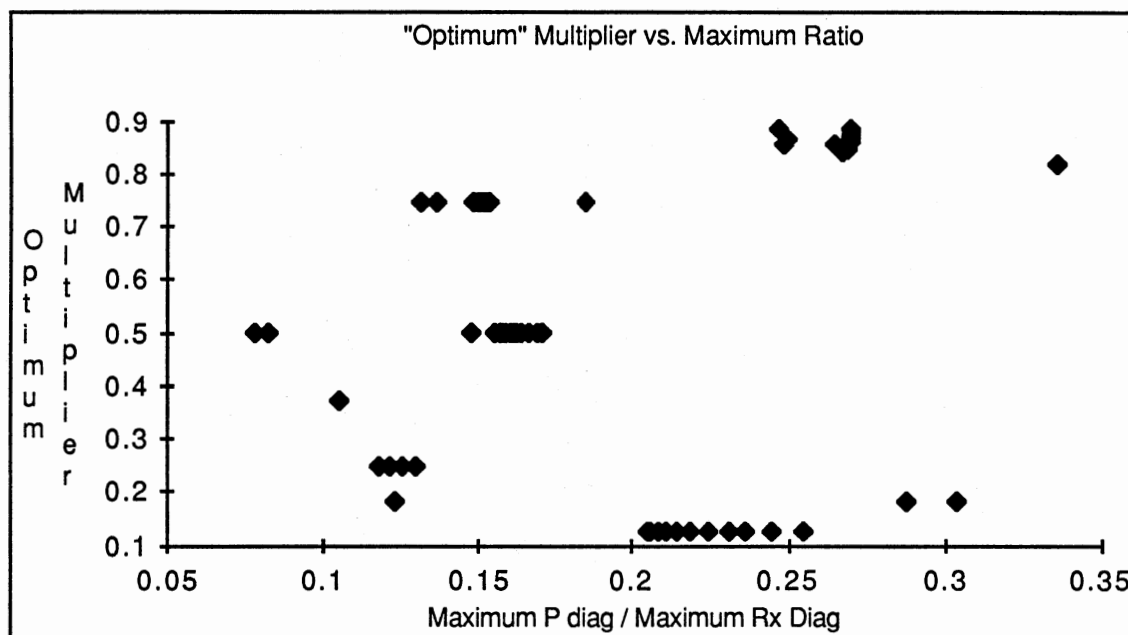


Figure 26. Optimum Multipliers

DOA Spectrum Peak Sharpness Measure

The L_p MUSIC algorithm is capable of determining directions of arrival of waves that are not resolved by the standard (L_2) MUSIC approach. When the

DOA spectrum for a particular value of p exhibits a peak that is not evident in the spectrum of another p value, it is clear that the former value of p is "better", at least as a detector of the wave that was missed by the latter p value. On the other hand, it is possible for a particular wave to produce peaks in the spectra for a variety of p values, in which case the decision of which p is "best" is somewhat more complicated.

One basis for comparison of competing values of p can be developed from the degree of sharpness exhibited by the various peaks in the DOA spectra under consideration. The degree of sharpness of a particular peak can be expressed as in Equation (4-1):

$$q(p) = \frac{1}{\Delta\theta} \log_{10} \sqrt[p]{f_p} \quad (4 - 1)$$

where $q(p)$ is the "sharpness" measure as a function of p , f_p is the magnitude ("height") of the peak being measured, and $\Delta\theta$ is the angular separation between the two points in the spectrum (one on either side of the peak) at which the spectrum value is some fixed increment below the peak value. The quantity f_p is the reciprocal of g_p from Equation (3-12); i.e. f_p is the "DOA Measure" that is plotted to obtain a DOA spectrum.

Equation (4-1) incorporates all the visual aspects of a particular peak, and so can be expected to provide an objective measure of the "sharpness" of the peak (which, of course, is a subjective trait). For instance, higher peaks produce larger values of q_p (assuming the other terms in Equation (4-1) don't change), since f_p is larger for higher peaks. Similarly, narrower peaks have larger q_p values, since $\Delta\theta$ is smaller for narrower peaks. Finally, taking the p^{th} root of the sum (of terms to the p^{th} power) which comprises f_p removes the direct effect of p variations, which is to produce larger f_p values for larger p values.

Figures 27 and 28 provide examples of the variation of q_p with p . The value for the "fixed increment" in these figures is 1db; i.e. $\Delta\theta$ is measured between the points at which the spectrum value is 1db below the peak value. The data used in generating Figures 27 and 28 was obtained from the experiment which produced Figures 19 through 22; with Figure 27 applicable before covariance matrix revision and Figure 28 applicable after covariance matrix revision. Figure 27 was produced using the spectrum peaks at about 35° in Figure 20, for p values ranging from 1 to 2.2. As can be seen from Figure 27, the peak sharpness measure decreases as resolution is lost (as p rises from 1 to 2.2), so a p value of 1 or 1.2 would be preferred for use in further estimation steps. Additionally, actual DOA error, as shown in the figure, generally rises as p increases from 1 to 2.2 and resolution is lost. Therefore, in this case selection of the p which produces the largest sharpness measure results in one of the best available DOA estimates (although this is not guaranteed in general).

Figure 28 was produced using the spectrum peaks at about 35° in Figure 21. As stated above, Figure 21 results from covariance matrix revision of the experiment which produced Figure 20. In the case of Figure 28, $p = 3$ provides the greatest sharpness measure, so would be preferred as the p value for use in subsequent estimation. A review of Figure 22 shows that the peak produced for $p = 3$ does appear to be the sharpest of the spectrum peaks in the neighborhood of 35° . Although in this case the actual DOA error (shown in Figure 28) is larger for $p = 3$ than for $p = 1$, the difference in DOA error is very slight across the range of p values.

Two other points of interest are evident from a comparison of Figure 27 with Figure 28. First, all the sharpness measures plotted in the upper sharpness curve of Figure 28 (after covariance revision) are larger than any of the sharpness measures shown in Figure 27 (before covariance matrix revision).

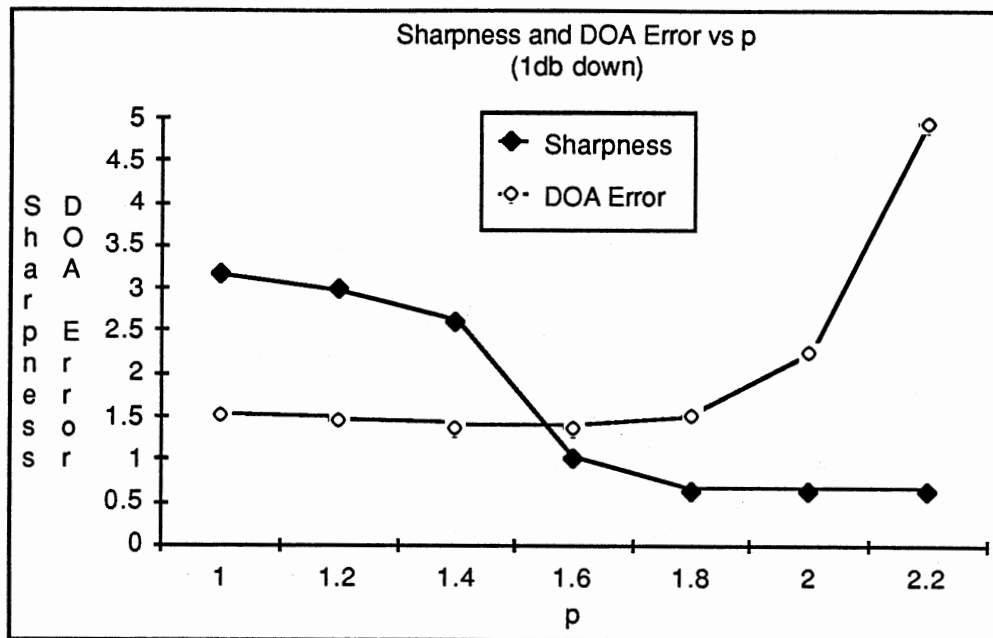


Figure 27. Sharpness Variation with p, Case 1

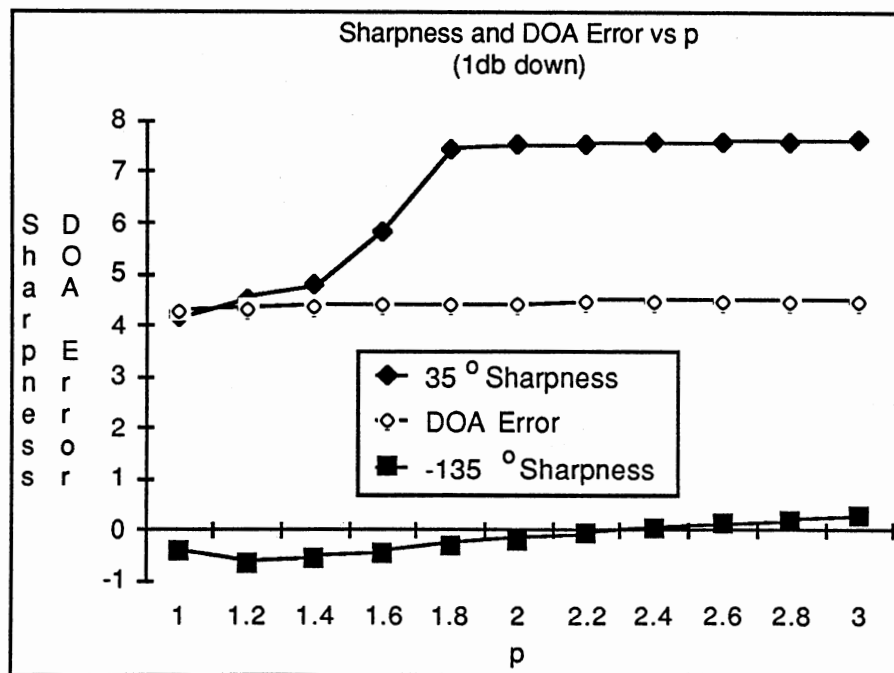


Figure 28. Sharpness Variation with p, Case 2

Thus, the sharpness measure accurately reflects the increase in peak sharpness produced by the iterative DOA revision approach.

The second point of interest is illustrated by comparing the upper sharpness curve with the lower sharpness curve in Figure 28. The upper curve is composed of the sharpness measures for the spectrum peaks at about 35° in Figure 22. The lower curve was produced for the spectrum peaks at about -135° , which do not correspond to an actual arriving wave. As is clear from Figure 28, the spurious peaks at -135° exhibit much lower sharpness values (and thus are much milder peaks) than do those at about 35° , which correspond to actual arriving waves. So, in this case a comparison of sharpness values assists not only in selection of a value of p but also in selection of which spectrum likely correspond to actual arriving waves.

Although the sharpest peak does not necessarily imply the most accurate DOA estimate (due to the complex interaction of the various influences on the DOA spectrum), q_p does provide a method for choosing one p value (or a collection of p values) over other values when problem constraints (e.g. available processing time) require such a choice. As mentioned above, q_p can also assist in selection of the DOA peaks which correspond to actual arriving waves, although again it is not guaranteed to do so in every case.

It is emphasized that Figures 27 and 28 are two examples out of many that may be produced. It is *not* true in general that $p = 1$ or $p = 3$ produces the best DOA estimates, nor that q_p is invariably highest for values of p which produce best DOA estimates. Simulations conducted during this research reveal that values of p other than two can, in some situations, produce superior resolution to that of standard (L_2) MUSIC, and that $p < 2$ is particularly useful in the presence of impulsive noise. On the other hand, the production of a DOA spectrum is an extremely complicated procedure, and it would be simplistic to claim

that any particular value of p (or of any other parameter used in L_p MUSIC) is best for all applications or even all possible data sets from a selected application.

CHAPTER V

CONCLUSIONS AND FUTURE RESEARCH

Conclusions

Two extensions were proposed to the MUSIC direction finding algorithm. The first, use of the L_p norm in the DOA spectrum expression (3-12), can provide improved DOA resolution in some situations, particularly where impulsive noise is present.

In the proposed L_p DOA approach, angles θ which minimize a sum (g_p) of terms to the p^{th} power are determined. These angles are L_p estimates of direction of arrival, in the sense that they minimize the p -norm of the difference between the array manifold ($\underline{a}(\theta)$) and the signal subspace (\underline{e}_n , $n = 1, 2, \dots, M$). The angle determination is accomplished by minimizing (in the p -norm sense) the projection of the vector $\underline{a}(\theta)$ onto the noise subspace. Since the noise subspace is the orthogonal complement of the signal subspace, this minimizes the distance between the array manifold and the signal subspace. That is, minimizing g_p selects the angles that best fit the observed data, in the p -norm sense, subject to the constraints of array geometry.

Additionally, no iteration is required to determine the L_p DOA estimates described above (as is necessary in the case of estimating \mathbf{P}). This implies that a \mathbf{P} estimate based on L_p DOA estimates (which can be considered an L_p estimate of \mathbf{P}) can be obtained with essentially no more computation than is required to obtain an L_2 estimate of \mathbf{P} (from the MUSIC DOA estimates). Also,

since no convergence issues are involved in minimizing g_p , one could obtain L_p DOA estimates (and the associated L_p \mathbf{P} estimate) for any value of $p > 0$ (although the interpretation for values of p outside the range $1 \leq p \leq 3$ might be open to question).

Further, since g_p provides a measure of the p -norm distance between the array manifold and the signal subspace, it gives at least a relative indication of the success of the estimation process. For example, if additional data is taken following an estimation step, and incorporation of the additional data results in lower values for g_p , it might be claimed that the estimates obtained using the additional data are in a sense "better" than the original estimates.

For the case of the number of arriving waves being one less than the number of antenna elements, the angles which minimize g_p are unchanged by varying p . This is because the sum to be minimized consists of one term for this case, and the minimum of a single term occurs at the same point regardless of the (positive) exponent to which that term is raised. Thus, when estimation of the directions of arrival of one fewer waves than array elements is attempted ("array saturation"), L_p DOA estimates are exactly the same as L_2 estimates, regardless of the (positive) value of p used. In this case the noise subspace consists of a line in N -space passing through the single noise eigenvector \underline{e}_N . Therefore, any local minimum of the projection of a vector (particularly the array manifold vector) onto the noise subspace (a line) can occur in only one direction, regardless of the norm used.

It was also possible to improve MUSIC's DOA resolution capability using an iterative DOA revision technique. The approach is based on a spectral subtraction of the effects of previously estimated directions of arrival from the L_p DOA spectrum. This technique was helpful whether the L_2 or the L_p norm was used, and provided the ability to resolve signals that are more closely spaced

than can be resolved with standard MUSIC. The technique involves empirical determination of a real multiplicative constant, and a numerical binary search procedure was proposed which permits determination of the constant's value to whatever resolution is desired. The search procedure allows for adjusting the resolution obtained based on application constraints such as computational burden.

Finally, an objective measure was developed of the sharpness of peaks in DOA spectra. This measure can be used to select which value or values of p are to be used during the estimation process, based on the variation in the sharpness measure over the range of p values.

Areas for Further Research

During the conduct of the research described herein, a number of issues arose which deserve further investigation. Two major areas for productive future research are alternatives to the algorithms used for DOA estimation, and implementation of these or other algorithms on multiple-processor computer systems.

Alternatives to MUSIC

The MUSIC algorithm was chosen as the basis for this research due to the fact that it is the most generally applicable of the high-resolution DOA estimation procedures. The L_p -norm DOA spectrum expression developed in Chapter III, however, could as easily have been developed for some other DOA estimation approach. For example, the sum-of-squares expressions used in the Maximum Likelihood, Burg, and Modified Forward-Backward Linear Predictor methods could be replaced by expressions involving sums of terms to the p^{th} power, with a possible improvement in DOA resolution similar to that shown by

the L_p MUSIC expression. Such revisions of existing techniques to accommodate L_p norm estimation could provide opportunities for improved performance with very little additional computational burden.

A related area of potential improvement involves the "array saturation" case described in Chapter III. In this situation, only one noise eigenvector is estimated, so that L_2 and L_p estimation produce the same results. The adaptive MUSIC method described in Chapter III, on the other hand, develops estimates of the signal subspace vectors rather than the noise subspace vectors. For the case of array saturation, then, it may be possible to apply the L_p norm to estimation of the signal subspace vectors, thereby possibly improving the resolution abilities of algorithms based on such estimates.

Use of Phase Information

The MUSIC algorithm employs the phase of the antenna array samples in the determination of directions of arrival. However, the IRLS procedure for estimating \mathbf{P} deals only with the magnitudes of the estimation residuals, thereby effectively ignoring potentially useful phase information. Use might be made of residual phase information, perhaps in an iterative procedure which alternately minimizes residual magnitudes and residual phases, to improve the performance of the overall estimation approach.

Also, the IRLS algorithm is by no means limited to powers of residual magnitudes as cost functions. A more general criterion function, which could include phase information, may produce a superior estimation algorithm. For example, $\ln(A) = |A| + j \arg(A)$, where A is a complex number, embodies both magnitude and phase information. It may be advantageous to use, for instance, some function of $\ln(r_i)$, where r_i is an estimation residual, as the cost function for IRLS processing.

Alternatives to IRLS

The IRLS algorithm is extremely computationally intensive. This is because a matrix pseudo-inverse is required at each IRLS iteration step (since the weight matrix changes at each step). An alternative approach, the Residual Steepest Descent (RSD) algorithm, can usually be employed to considerably reduce the computational burden. The RSD algorithm requires the matrix inversion only once, at the beginning of the estimation process. Although the RSD algorithm does not guarantee convergence to an exact L_p solution under all circumstances, it is accurate enough for use in most situations. Therefore, an RSD based approach might provide the advantages of L_p estimation with a reduced computational load.

For the case of L_1 estimation, linear programming approaches are available; however, they are not very attractive (Lansford, 1985). Direct implementation of the simplex linear programming solution is usually prohibitive in terms of memory required; however, Karmarkar's (1984) algorithm, among others, can be used to circumvent this difficulty to some extent. Use of such linear programming algorithms for L_1 estimation should therefore be investigated.

Adaptation to Multiple Processor Systems

The simulation program mentioned in Chapter III was developed for a general purpose, single-processor system. Adaptation for multiple-processor computers of the various algorithms discussed can be expected to provide the opportunity for improved performance. Parallel implementations of some of the procedures required by MUSIC and the other estimation routines already exist

(e.g. matrix multiplication and singular value decomposition). Of course, parallel implementation of a complete DOA estimation system requires more than merely "stringing together" the parallel versions of the various routines employed, and is a subject worthy of intense study.

As mentioned in Chapter III, Van Loan's generalized singular value decomposition was used to solve the generalized eigenvalue problem of Chapter II. Although a parallel version of Van Loan's algorithm may be possible, another algorithm, due to Paige (1986), is available for computing the generalized singular value decomposition. A parallel version of this algorithm has already been developed (Heath, et. al., 1986).

Finally, the iterative DOA revision process described in Chapter III, although not apparently a ready candidate for parallel processing, may be well suited to a Multiple Instruction stream Multiple Data stream (MIMD) architecture. Research currently in progress at Oklahoma State University is directed toward exploiting the exceptional power of this architecture, as applied to problems involving multiple similar but not identical processes.

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APPENDIX A

CONVERGENCE OF THE ITERATIVELY REWEIGHTED LEAST SQUARES (IRLS) ALGORITHM

The objective of the IRLS algorithm is to estimate a vector \underline{f} given the linear model:

$$\underline{x} = \mathbf{A}\underline{f} + \underline{w};$$

where \underline{x} is an N-element output vector, \mathbf{A} is an NxM matrix of rank M, \underline{f} is an M-element vector of unknown parameters, and \underline{w} is an N-element zero-mean noise vector. The IRLS algorithm repeatedly evaluates:

$$\hat{\underline{f}}_{t+1} = [\mathbf{A}^H \mathbf{W}(\hat{\underline{f}}_t) \mathbf{A}]^{-1} \mathbf{A}^H \mathbf{W}(\hat{\underline{f}}_t) \underline{x} \quad (\text{A} - 1)$$

until $\hat{\underline{f}}$ converges. The weighting matrix \mathbf{W} is a diagonal matrix with entries:

$$w_{nn} = w_n = w(r_n) = \psi(r_n)/r_n = \rho'(r_n)/r_n \quad (\text{A} - 2)$$

where $r_n = \underline{a}_n \hat{\underline{f}} - x_n$; \underline{a}_n is the n^{th} row of \mathbf{A} ; x_n is the n^{th} element of \underline{x} ; and the objective function ρ is a positive, even, differentiable function. This definition of \mathbf{W} results in minimization of the loss function J, as defined below:

$$J = \sum_{n=1}^N \rho(\underline{a}_n \hat{\underline{f}} - x_n) \quad (\text{A} - 3)$$

That iteration of Equation (A-1) using \mathbf{W} as defined in Equation (A-2) minimizes J, above, can be seen as follows.

If $\rho(r)$ is nondecreasing in $|r|$, then for J to be minimized, it is necessary only that:

$$\frac{\partial J}{\partial f_m} = \sum_{n=1}^N \rho'(\mathbf{a}_n \mathbf{f} - x_n) a_{nm} = 0; \quad m = 1, 2, \dots, M$$

where f_m is the m^{th} element of \mathbf{f} , and a_{nm} is the m^{th} element of the n^{th} row of \mathbf{A} .

In vector form:

$$\mathbf{A}^T [\rho'(r_n)] = \mathbf{0} \quad (\text{A-4})$$

where $[\rho'(r_n)]$ is the N -element vector of derivatives evaluated at r_n ; $n = 1, 2, \dots, N$, and $\mathbf{A}^T [\rho'(r_n)]$ is an M -vector. Since $w(r_n) = \rho'(r_n)/r_n$:

$$\mathbf{W}(r_n) [r_n] = [\rho'(r_n)]$$

where $\mathbf{W}(r_n)$ is a diagonal matrix with elements $w(r_n)$ and $[r_n]$ is the N -element vector of residuals r_n . So, from Equation (A-4), we have:

$$\mathbf{A}^T \mathbf{W}(r_n) [r_n] = \mathbf{0} \quad (\text{A-5})$$

where $\mathbf{A}^T \mathbf{W}(r_n) [r_n]$ is an M -vector.

Now consider the change in \mathbf{f} at iteration step t , given by the difference $\hat{\mathbf{f}}_{t+1} - \hat{\mathbf{f}}_t$. Using Equations (A-1) and (A-2) gives:

$$\begin{aligned} \hat{\mathbf{f}}_{t+1} - \hat{\mathbf{f}}_t &= \{\mathbf{A}^T \mathbf{W}(r_n) \mathbf{A}\}^{-1} \mathbf{A}^T \mathbf{W}(r_n) \mathbf{z} - \hat{\mathbf{f}}_t \\ &= \{\mathbf{A}^T \mathbf{W}(r_n) \mathbf{A}\}^{-1} \mathbf{A}^T \mathbf{W}(r_n) \mathbf{z} - \{\mathbf{A}^T \mathbf{W}(r_n) \mathbf{A}\}^{-1} \{\mathbf{A}^T \mathbf{W}(r_n) \mathbf{A}\} \hat{\mathbf{f}}_t \\ &= -\{\mathbf{A}^T \mathbf{W}(r_n) \mathbf{A}\}^{-1} \mathbf{A}^T \mathbf{W}(r_n) \{\mathbf{A} \hat{\mathbf{f}}_t - \mathbf{z}\} \end{aligned}$$

Now when \mathbf{f} converges, $\hat{\mathbf{f}}_{t+1} - \hat{\mathbf{f}}_t \rightarrow \mathbf{0}$. Thus, since $[r_n] = \mathbf{A} \hat{\mathbf{f}}_t - \mathbf{z}$:

$$\hat{\mathbf{f}}_{t+1} - \hat{\mathbf{f}}_t = -\{\mathbf{A}^T \mathbf{W}(r_n) \mathbf{A}\}^{-1} \mathbf{A}^T \mathbf{W}(r_n) [r_n] \rightarrow \mathbf{0}$$

implies that $\mathbf{A}^T \mathbf{W}(r_n) [r_n] \rightarrow \mathbf{0}$, which, from Equation (A-5), means that J is minimized.

This means that use of Equation (A-1), with \mathbf{W} as defined in Equation (A-2), minimizes the cost function given in Equation (A-3) if it (Equation (A-1)) converges. Global convergence of the algorithm is guaranteed for objective functions ρ such that:

- (1) $\rho(r)$ is a differentiable, symmetric, positive function nondecreasing in $|r|$,
- (2) $\rho(r) \rightarrow \infty$ as $|r| \rightarrow \infty$,
- (3) $w(r) = \rho'(r)/r$ is nonincreasing in $|r|$, and
- (4) $w(r) = \rho'(r)/r$ is bounded for all r .

For L_p estimation, the required objective function is:

$$\begin{aligned} \rho(r) &= |r|^p; |r| > \varepsilon \\ &= |r|^p; |r| \leq \varepsilon; \varepsilon \text{ some small positive number} \end{aligned}$$

Since this function meets conditions (1) - (4) for values of p such that $1 \leq p \leq 2$, global convergence of the algorithm is guaranteed for p in this range. Additionally, local convergence of the algorithm is guaranteed for the above function for $2 < p \leq 3$. For $p > 3$, the algorithm diverges (Byrd and Pyne, 1979).

APPENDIX B

THE GENERALIZED SINGULAR VALUE DECOMPOSITION

The MUSIC algorithm requires the solution to a generalized eigenvalue problem involving two covariance matrices. That is, estimates are sought of the generalized eigenvalues and eigenvectors of the matrix pair $(\mathbf{R}_x, \mathbf{R}_b)$, associated with a collection of S signal samples \underline{x}_s and S noise samples \underline{w}_s . The signal and noise samples are generally "snapshots", or samples at various instants, from an antenna array. The matrices \mathbf{R}_x and \mathbf{R}_b are given by:

$$\mathbf{R}_x \equiv E\{ \underline{x} \underline{x}^H \} \quad (\text{B} - 1)$$

$$\lambda_{\min} \mathbf{R}_b \equiv E\{ \underline{w} \underline{w}^H \} \quad (\text{B} - 2)$$

where λ_{\min} normalizes \mathbf{R}_b such that $\text{tr}\{ \mathbf{R}_b \} = N$, and $E\{\}$ denotes statistical expectation over s .

In practice, direct solution of a generalized eigenvalue problem involving matrices of the form $\mathbf{M}^H \mathbf{M}$ or $\mathbf{M} \mathbf{M}^H$ is not only computationally intensive but also can lead to numerical difficulties. This is due to the fact that the condition number of either matrix product is the square of the condition number of the original matrix (\mathbf{M}). It is therefore advantageous to devise a method whereby the generalized eigenvalue problem can be solved without requiring the actual formation of the matrix products listed. One approach to this problem is via a Generalized Singular Value Decomposition (GSVD) algorithm (Speiser, 1985).

A GSVD algorithm applied to the matrix pair (\mathbf{A}, \mathbf{B}) computes matrices \mathbf{U} , \mathbf{Q} , \mathbf{Z} , \mathbf{D}_A , and \mathbf{D}_B such that:

$$\mathbf{U}^H \mathbf{A} \mathbf{Z} = \mathbf{D}_A \quad (\text{B-3})$$

$$\mathbf{Q}^H \mathbf{B} \mathbf{Z} = \mathbf{D}_B \quad (\text{B-4})$$

where:

\mathbf{U} and \mathbf{Q} are unitary ($\mathbf{U}^H \mathbf{U} = \mathbf{I}$, $\mathbf{Q}^H \mathbf{Q} = \mathbf{I}$),

\mathbf{Z} is non-singular, and

\mathbf{D}_A and \mathbf{D}_B are real and diagonal.

If the conjugate transposes of Equations (B-3) and (B-4) are evaluated and postmultiplied by their original forms, the following are obtained:

$$\mathbf{Z}^H \mathbf{A}^H \mathbf{A} \mathbf{Z} = (\mathbf{D}_A)^2 \quad (\text{B-5})$$

$$\mathbf{Z}^H \mathbf{B}^H \mathbf{B} \mathbf{Z} = (\mathbf{D}_B)^2 \quad (\text{B-6})$$

Now, multiplying Equation (B-6) by λ_n and subtracting it from Equation (B-5) gives:

$$\mathbf{Z}^H (\mathbf{A}^H \mathbf{A} - \lambda_n \mathbf{B}^H \mathbf{B}) \mathbf{Z} = (\mathbf{D}_A)^2 - \lambda_n (\mathbf{D}_B)^2 \quad (\text{B-7})$$

Evaluating Equation (B-7) with $\lambda_n = d_{A,n}^2 / d_{B,n}^2$, where $d_{A,n}$ and $d_{B,n}$ are the n^{th} elements of \mathbf{D}_A and \mathbf{D}_B , respectively, produces a singular matrix on the right-hand side. This is true because the difference on the right-hand side of Equation (B-7) is diagonal, with a zero entry at the n^{th} (diagonal) element. On the left-hand side, since \mathbf{Z} is non-singular, the expression in parenthesis must be singular. Thus:

$$|\mathbf{A}^H \mathbf{A} - \lambda_n \mathbf{B}^H \mathbf{B}| = 0 \quad (\text{B-8})$$

This can be true (for non-singular $\mathbf{A}^H \mathbf{A}$ and $\mathbf{B}^H \mathbf{B}$) only for λ_n a generalized eigenvalue of the matrix pair $(\mathbf{A}^H \mathbf{A}, \mathbf{B}^H \mathbf{B})$. Thus, the generalized eigenvalues of the matrix pair $(\mathbf{A}^H \mathbf{A}, \mathbf{B}^H \mathbf{B})$ are given by the ratios of the squares of the singular values $d_{A,n}$ and $d_{B,m}$, above.

Now consider the n^{th} diagonal entry in Equation (B-7):

$$\mathbf{z}_n^H (\mathbf{A}^H \mathbf{A} - \lambda_n \mathbf{B}^H \mathbf{B}) \mathbf{z}_n = d_{A,n}^2 - \lambda_n d_{B,n}^2 \quad (\text{B-9})$$

If this equation is evaluated with $\lambda_n = d_{A,n}^2 / d_{B,n}^2$, the result is 0. This indicates that the generalized eigenvectors of the matrix pair $(\mathbf{A}^H\mathbf{A}, \mathbf{B}^H\mathbf{B})$ are the columns of \mathbf{Z} .

Thus it is clear that an algorithm which computes a generalized SVD can also be used to produce the generalized eigenvalues and eigenvectors of a matrix pair of the form described above.

It is, in fact, not even necessary to deal with the matrices \mathbf{A} and \mathbf{B} directly, if this is not desired. Such a situation could occur, for instance, if the matrices of interest had a very large number of rows but relatively few columns. In such a case it would be preferable to deal with matrices whose size was strictly determined by the number of columns, rather than the number of rows, in the matrices of interest.

Consider an $S \times N$ matrix \mathbf{A} of rank N with $S \geq N$ (that is, the columns of \mathbf{A} are linearly independent). A QR decomposition can be applied to such a matrix to produce:

$$\mathbf{A}_{S \times N} = \mathbf{Q}_{S \times N} \mathbf{T}_{N \times N} \quad (\text{B} - 10)$$

where the columns of \mathbf{Q} are orthonormal and \mathbf{T} is upper triangular and invertible. Evaluating $\mathbf{A}^H\mathbf{A}$ gives:

$$\mathbf{A}^H\mathbf{A} = \mathbf{T}^H\mathbf{Q}^H\mathbf{Q}\mathbf{T} = \mathbf{T}^H\mathbf{T} \quad (\text{B} - 11)$$

(since $\mathbf{Q}^H\mathbf{Q}$ is an $N \times N$ identity matrix). Thus, using \mathbf{T} in the generalized singular value decomposition is exactly equivalent to using the full \mathbf{A} matrix (as far as concerns the generalized eigenvalue problem).

So it is evident that, except for an initial QR decomposition step, it is possible to deal exclusively with comparatively small matrices and still solve the generalized eigenvalue problem for the matrix pair $(\mathbf{A}^H\mathbf{A}, \mathbf{B}^H\mathbf{B})$ without forming the actual matrix products.

To apply the above approach to the matrices used in the MUSIC algorithm, it is only necessary to form the matrices \mathbf{X} and \mathbf{W} from the signal and noise vector samples as follows:

$$\mathbf{X} \equiv \frac{1}{\sqrt{I}} \begin{bmatrix} \mathbf{x}_1^H \\ \mathbf{x}_2^H \\ \vdots \\ \mathbf{x}_I^H \end{bmatrix}; \mathbf{W} \equiv \frac{1}{\sqrt{I\lambda_{\min}}} \begin{bmatrix} \mathbf{w}_1^H \\ \mathbf{w}_2^H \\ \vdots \\ \mathbf{w}_I^H \end{bmatrix} \quad (\text{B} - 12)$$

Application of a generalized SVD approach as described above (either with or without a preliminary QR decomposition step) determines the generalized eigenvalues and eigenvectors of the matrix pair $(\mathbf{X}^H\mathbf{X}, \mathbf{W}^H\mathbf{W})$. From Equation (B-12) it can be seen that:

$$\mathbf{X}^H\mathbf{X} = \frac{1}{S} \sum_{s=1}^S \mathbf{x}_s \mathbf{x}_s^H \quad (\text{B} - 13)$$

$$\lambda_{\min} \mathbf{W}^H\mathbf{W} = \frac{1}{S} \sum_{s=1}^S \mathbf{w}_s \mathbf{w}_s^H \quad (\text{B} - 14)$$

That is, $\mathbf{X}^H\mathbf{X} \approx \mathbf{R}_x$ and $\mathbf{W}^H\mathbf{W} \approx \mathbf{R}_b$. Thus, the approximations desired can be obtained using a generalized SVD approach.

At least two generalized SVD algorithms are available, a comparatively older one due to Van Loan and a more recent one by Paige (Van Loan, 1985) (Paige, 1986). Van Loan's approach, when applied following a QR decomposition, has the possible advantage of producing an identity matrix for \mathbf{D}_A . However, it also involves an explicit matrix inversion, which could lead to numerical difficulty. Although the approach due to Paige is less straightforward, it does not involve any explicit matrix inversions, and so is likely to be the more robust algorithm. Also, a parallel implementation of Paige's algorithm has been de-

veloped by Heath, et. al. (1986). Paige's method uses a preliminary QR decomposition step.

Van Loan's approach was implemented in the simulation program described in Appendix D. It can be applied to the MUSIC algorithm's generalized eigenvalue problem using the following steps:

- 1) Perform QR decompositions of \mathbf{X} and \mathbf{W} (defined above):

$$\mathbf{X} = \mathbf{L}_X \mathbf{T}_X; \mathbf{L}_X^H \mathbf{L}_X = \mathbf{I}; \mathbf{T}_X \text{ upper triangular, invertible.}$$

$$\mathbf{W} = \mathbf{L}_W \mathbf{T}_W; \mathbf{L}_W^H \mathbf{L}_W = \mathbf{I}; \mathbf{T}_W \text{ upper triangular, invertible.}$$

- 2) Compute (normal) singular value decomposition of $\mathbf{T}_W \mathbf{T}_X^{-1}$:

$$\mathbf{T}_W \mathbf{T}_X^{-1} = \mathbf{Q} \mathbf{D}_B \mathbf{U}^H; \mathbf{Q}^H \mathbf{Q} = \mathbf{I}; \mathbf{U}^H \mathbf{U} = \mathbf{I}; \mathbf{D}_B \text{ real, diagonal.}$$

- 3) Set $\mathbf{Z} = \mathbf{T}_X^{-1} \mathbf{U}$.

The matrices \mathbf{U} , \mathbf{Q} , and \mathbf{Z} computed above perform the desired generalized singular value decomposition, which can be seen as follows:

$$\mathbf{U}^H \mathbf{T}_X \mathbf{Z} = \mathbf{U}^H \mathbf{T}_X \mathbf{T}_X^{-1} \mathbf{U} = \mathbf{U}^H \mathbf{U} = \mathbf{I} \text{ (which is real \& diagonal).}$$

$$\mathbf{Q}^H \mathbf{T}_W \mathbf{Z} = \mathbf{Q}^H (\mathbf{T}_W \mathbf{T}_X^{-1}) \mathbf{U} = \mathbf{Q}^H (\mathbf{Q} \mathbf{D}_B \mathbf{U}^H) \mathbf{U} = (\mathbf{Q}^H \mathbf{Q}) \mathbf{D}_B (\mathbf{U}^H \mathbf{U}) = \mathbf{D}_B$$

Thus, the computed \mathbf{U} , \mathbf{Q} , and \mathbf{Z} comprise a generalized singular value decomposition of the triangular matrices \mathbf{T}_X and \mathbf{T}_W . This, as shown above, suffices to determine the generalized eigenvalues and eigenvectors of the matrix product pair $(\mathbf{X}^H \mathbf{X}, \mathbf{W}^H \mathbf{W})$, which is the desired result.

APPENDIX C

THE MINIMUM NORM ESTIMATE OF A VECTOR

In applying the L_p norm to the estimation problems addressed in this dissertation, it becomes necessary in several situations to evaluate expressions of the form $(\mathbf{A}^H \mathbf{W} \mathbf{A})^{-1} \mathbf{A}^H \mathbf{W}$, where \mathbf{W} is real, nonsingular, and diagonal. In order to avoid an explicit matrix inversion in the evaluation of the above expression, a procedure based on the Moore-Penrose pseudo-inverse is used. The approach proceeds as follows:

$$(\mathbf{A}^H \mathbf{W} \mathbf{A})^{-1} \mathbf{A}^H \mathbf{W} = \left(\mathbf{A}^H \mathbf{W}^{\frac{1}{2}} \mathbf{W}^{\frac{1}{2}} \mathbf{A} \right)^{-1} \mathbf{A}^H \mathbf{W}^{\frac{1}{2}} \mathbf{W}^{\frac{1}{2}}$$

and, since \mathbf{W} is real and diagonal:

$$(\mathbf{A}^H \mathbf{W} \mathbf{A})^{-1} \mathbf{A}^H \mathbf{W} = \left(\left(\mathbf{W}^{\frac{1}{2}} \mathbf{A} \right)^H \mathbf{W}^{\frac{1}{2}} \mathbf{A} \right)^{-1} \left(\mathbf{W}^{\frac{1}{2}} \mathbf{A} \right)^H \mathbf{W}^{\frac{1}{2}}$$

Defining:

$$\mathbf{B} \equiv \left(\mathbf{W}^{\frac{1}{2}} \mathbf{A} \right)^H$$

gives:

$$(\mathbf{A}^H \mathbf{W} \mathbf{A})^{-1} \mathbf{A}^H \mathbf{W} = (\mathbf{B}^H \mathbf{B})^{-1} \mathbf{B}^H \mathbf{W}^{\frac{1}{2}}$$

Now $(\mathbf{B}^H \mathbf{B})^{-1} \mathbf{B}^H$ is the Moore-Penrose pseudo-inverse of \mathbf{B} , and computing the square root of \mathbf{W} requires little more work than would computing \mathbf{W} itself, since the elements of \mathbf{W} are generally exponentials. This implies that, except

for the addition of one initial and one final diagonal matrix multiply at each iteration, the same operations can be used to form L_p estimates as are used to form L_2 estimates.

Actual evaluation of the Moore-Penrose pseudo-inverse of \mathbf{B} is accomplished using an algorithm based on the singular value decomposition. Although proofs that such an algorithm produces minimum-norm vector estimates are fairly common for the real case, the more general case for complex vectors and matrices is covered only rarely (Graybill, 1969). Therefore, a proof for complex data is included here.

Problem Statement

It is desired to find the minimum (L_2) norm solution to the equation:

$$\mathbf{Ax} = \mathbf{b}$$

where \mathbf{A} is $N \times M$, $N \geq M$, of rank $r \leq M$. That is, the solution $\underline{x} = \underline{x}_0$ is sought such that:

$$(\mathbf{Ax} - \mathbf{b})^H(\mathbf{Ax} - \mathbf{b}) \geq (\mathbf{Ax}_0 - \mathbf{b})^H(\mathbf{Ax}_0 - \mathbf{b}), \text{ all } \underline{x} \text{ in } \mathbb{C}^N; \text{ and}$$

$$\underline{x}^H \underline{x} > \underline{x}_0^H \underline{x}_0 \text{ for } \underline{x} \neq \underline{x}_0 \text{ and } (\mathbf{Ax} - \mathbf{b})^H(\mathbf{Ax} - \mathbf{b}) = (\mathbf{Ax}_0 - \mathbf{b})^H(\mathbf{Ax}_0 - \mathbf{b}).$$

Solution

The singular value decomposition of \mathbf{A} yields $\mathbf{A} = \mathbf{U}\mathbf{\Sigma}\mathbf{V}^H$ with \mathbf{U} $N \times N$, $\mathbf{U}^H\mathbf{U} = \mathbf{U}\mathbf{U}^H = \mathbf{I}_{N \times N}$; \mathbf{V} $M \times M$, $\mathbf{V}^H\mathbf{V} = \mathbf{V}\mathbf{V}^H = \mathbf{I}_{M \times M}$; and $\mathbf{\Sigma}$ $N \times M$, real, diagonal, as follows:

$$\begin{bmatrix} \sigma_1 & 0 & \dots & \dots & \dots & \dots & 0 \\ 0 & \sigma_2 & \ddots & & & & \vdots \\ \vdots & \ddots & \ddots & \ddots & & & \vdots \\ \vdots & & \ddots & \sigma_r & \ddots & & \vdots \\ \vdots & & & \ddots & 0 & \ddots & \vdots \\ \vdots & & & & \ddots & \ddots & 0 \\ \vdots & & & & & \ddots & 0 \\ \vdots & & & & & & 0 \\ \vdots & & & & & & \vdots \\ 0 & \dots & \dots & \dots & \dots & \dots & 0 \end{bmatrix}$$

The desired minimum norm solution is $\underline{x}_0 = \mathbf{A}^- \underline{b}$, where $\mathbf{A}^- = \mathbf{V}\mathbf{X}^T\mathbf{U}^H$,
and the $N \times M$ matrix \mathbf{X} is:

$$\begin{bmatrix} \sigma_1^{-1} & 0 & \dots & \dots & \dots & \dots & 0 \\ 0 & \sigma_2^{-1} & \ddots & & & & \vdots \\ \vdots & \ddots & \ddots & \ddots & & & \vdots \\ \vdots & & \ddots & \sigma_r^{-1} & \ddots & & \vdots \\ \vdots & & & \ddots & 0 & \ddots & \vdots \\ \vdots & & & & \ddots & \ddots & 0 \\ \vdots & & & & & \ddots & 0 \\ \vdots & & & & & & 0 \\ \vdots & & & & & & \vdots \\ 0 & \dots & \dots & \dots & \dots & \dots & 0 \end{bmatrix}$$

Proof

First, the proposed \mathbf{A}^- will be shown to have the following properties:

- 1) $\mathbf{A}\mathbf{A}^-$ is Hermitian,
- 2) $\mathbf{A}^-\mathbf{A}$ is Hermitian,
- 3) $\mathbf{A}\mathbf{A}^-\mathbf{A} = \mathbf{A}$
- 4) $\mathbf{A}^-\mathbf{A}\mathbf{A}^- = \mathbf{A}^-$

To demonstrate the above properties, note that:

$\Sigma X^T = (\Sigma X^T)^T = N \times N$ real diagonal matrix containing r ones:

$$\begin{bmatrix} 1_{(1)} & 0 & \dots & \dots & \dots & \dots & 0 \\ 0 & 1_{(2)} & \ddots & & & & \vdots \\ \vdots & \ddots & \ddots & \ddots & & & \vdots \\ \vdots & & \ddots & 1_{(r)} & \ddots & & \vdots \\ \vdots & & & \ddots & 0 & \ddots & \vdots \\ \vdots & & & & \ddots & \ddots & 0 \\ 0 & \dots & \dots & \dots & \dots & 0 & 0_{(N)} \end{bmatrix}$$

$X^T \Sigma = (X^T \Sigma)^T = M \times M$ real diagonal matrix containing r ones:

$$\begin{bmatrix} 1_{(1)} & 0 & \dots & \dots & \dots & \dots & 0 \\ 0 & 1_{(2)} & \ddots & & & & \vdots \\ \vdots & \ddots & \ddots & \ddots & & & \vdots \\ \vdots & & \ddots & 1_{(r)} & \ddots & & \vdots \\ \vdots & & & \ddots & 0 & \ddots & \vdots \\ \vdots & & & & \ddots & \ddots & 0 \\ 0 & \dots & \dots & \dots & \dots & 0 & 0_{(M)} \end{bmatrix}$$

$$\Sigma (X^T \Sigma) = \Sigma; (X^T \Sigma) X^T = X^T.$$

To prove property (1):

$$\begin{aligned} [AA^{-}]^H &= [(U\Sigma V^H)(VX^T U^H)]^H \\ &= [U\Sigma V^H V X^T U^H]^H \\ &= [U(\Sigma X^T) U^H]^H \\ &= U(\Sigma X^T)^T U^H \\ &= AA^{-} \end{aligned}$$

To prove property (2):

$$\begin{aligned} [A^{-}A]^H &= [(VX^T U^H)(U\Sigma V^H)]^H \\ &= [VX^T U^H U \Sigma V^H]^H \\ &= [V(X^T \Sigma) V^H]^H \end{aligned}$$

$$\begin{aligned}
&= \mathbf{V}(\mathbf{X}^T \boldsymbol{\Sigma})^T \mathbf{V}^H \\
&= \mathbf{A}^* \mathbf{A}
\end{aligned}$$

To prove property (3):

$$\begin{aligned}
\mathbf{A} \mathbf{A}^* \mathbf{A} &= \mathbf{U} \boldsymbol{\Sigma} \mathbf{X}^T \mathbf{U}^H \mathbf{U} \boldsymbol{\Sigma} \mathbf{V}^H \\
&= \mathbf{U} \boldsymbol{\Sigma} (\mathbf{X}^T \boldsymbol{\Sigma}) \mathbf{V}^H \\
&= \mathbf{U} \boldsymbol{\Sigma} \mathbf{V}^H \\
&= \mathbf{A}
\end{aligned}$$

To prove property (4):

$$\begin{aligned}
\mathbf{A}^* \mathbf{A} \mathbf{A}^* &= \mathbf{V} \mathbf{X}^T \boldsymbol{\Sigma} \mathbf{V}^H \mathbf{V} \mathbf{X}^T \mathbf{U}^H \\
&= \mathbf{V} (\mathbf{X}^T \boldsymbol{\Sigma}) \mathbf{X}^T \mathbf{U}^H \\
&= \mathbf{V} \mathbf{X}^T \mathbf{U}^H \\
&= \mathbf{A}^*
\end{aligned}$$

Properties (1) - (4) can be used to prove that the proposed solution is the minimum norm solution as follows:

$$\begin{aligned}
(\mathbf{A} \mathbf{x} - \mathbf{b})^H (\mathbf{A} \mathbf{x} - \mathbf{b}) &= (\mathbf{A} \mathbf{x} - \mathbf{A} \mathbf{A}^* \mathbf{b} + \mathbf{A} \mathbf{A}^* \mathbf{b} - \mathbf{b})^H (\mathbf{A} \mathbf{x} - \mathbf{A} \mathbf{A}^* \mathbf{b} + \mathbf{A} \mathbf{A}^* \mathbf{b} - \mathbf{b}) \\
&= [\mathbf{A}(\mathbf{x} - \mathbf{A}^* \mathbf{b}) + (\mathbf{A} \mathbf{A}^* - \mathbf{I}) \mathbf{b}]^H [\mathbf{A}(\mathbf{x} - \mathbf{A}^* \mathbf{b}) + (\mathbf{A} \mathbf{A}^* - \mathbf{I}) \mathbf{b}] \\
&= [\mathbf{A}(\mathbf{x} - \mathbf{A}^* \mathbf{b})]^H [\mathbf{A}(\mathbf{x} - \mathbf{A}^* \mathbf{b})] + [\mathbf{A}(\mathbf{x} - \mathbf{A}^* \mathbf{b})]^H [(\mathbf{A} \mathbf{A}^* - \mathbf{I}) \mathbf{b}] \\
&\quad + [(\mathbf{A} \mathbf{A}^* - \mathbf{I}) \mathbf{b}]^H [\mathbf{A}(\mathbf{x} - \mathbf{A}^* \mathbf{b})] + [(\mathbf{A} \mathbf{A}^* - \mathbf{I}) \mathbf{b}]^H [(\mathbf{A} \mathbf{A}^* - \mathbf{I}) \mathbf{b}] \\
&= [\mathbf{A}(\mathbf{x} - \mathbf{A}^* \mathbf{b})]^H [\mathbf{A}(\mathbf{x} - \mathbf{A}^* \mathbf{b})] + [(\mathbf{A} \mathbf{A}^* - \mathbf{I}) \mathbf{b}]^H [(\mathbf{A} \mathbf{A}^* - \mathbf{I}) \mathbf{b}] \\
&\quad + (\mathbf{A} \mathbf{A}^* \mathbf{b} - \mathbf{I} \mathbf{b})^H (\mathbf{A} \mathbf{x} - \mathbf{A} \mathbf{A}^* \mathbf{b}) + (\mathbf{A} \mathbf{x} - \mathbf{A} \mathbf{A}^* \mathbf{b})^H (\mathbf{A} \mathbf{A}^* \mathbf{b} - \mathbf{I} \mathbf{b}) \\
&= [\mathbf{A}(\mathbf{x} - \mathbf{A}^* \mathbf{b})]^H [\mathbf{A}(\mathbf{x} - \mathbf{A}^* \mathbf{b})] + [(\mathbf{A} \mathbf{A}^* - \mathbf{I}) \mathbf{b}]^H [(\mathbf{A} \mathbf{A}^* - \mathbf{I}) \mathbf{b}] \\
&\quad + (\mathbf{A} \mathbf{A}^* \mathbf{b})^H \mathbf{A} \mathbf{x} - (\mathbf{A} \mathbf{A}^* \mathbf{b})^H \mathbf{A} \mathbf{A}^* \mathbf{b} - \mathbf{b}^H \mathbf{A} \mathbf{x} + \mathbf{b}^H \mathbf{A} \mathbf{A}^* \mathbf{b} \\
&\quad + (\mathbf{A} \mathbf{x})^H \mathbf{A} \mathbf{A}^* \mathbf{b} - (\mathbf{A} \mathbf{x})^H \mathbf{b} - (\mathbf{A} \mathbf{A}^* \mathbf{b})^H \mathbf{A} \mathbf{A}^* \mathbf{b} + (\mathbf{A} \mathbf{A}^* \mathbf{b})^H \mathbf{b} \\
&= [\mathbf{A}(\mathbf{x} - \mathbf{A}^* \mathbf{b})]^H [\mathbf{A}(\mathbf{x} - \mathbf{A}^* \mathbf{b})] + [(\mathbf{A} \mathbf{A}^* - \mathbf{I}) \mathbf{b}]^H [(\mathbf{A} \mathbf{A}^* - \mathbf{I}) \mathbf{b}] \\
&\quad + \mathbf{b}^H (\mathbf{A} \mathbf{A}^*)^H \mathbf{A} \mathbf{x} - \mathbf{b}^H (\mathbf{A} \mathbf{A}^*)^H \mathbf{A} \mathbf{A}^* \mathbf{b} - \mathbf{b}^H \mathbf{A} \mathbf{x} + \mathbf{b}^H \mathbf{A} \mathbf{A}^* \mathbf{b} \\
&\quad + \mathbf{x}^H \mathbf{A}^H \mathbf{A} \mathbf{A}^* \mathbf{b} - \mathbf{x}^H \mathbf{A}^H \mathbf{b} - \mathbf{b}^H (\mathbf{A} \mathbf{A}^*)^H \mathbf{A} \mathbf{A}^* \mathbf{b} + \mathbf{b}^H (\mathbf{A} \mathbf{A}^*)^H \mathbf{b}
\end{aligned}$$

$$\begin{aligned}
&= (\mathbf{A}^{-}\underline{\mathbf{b}})^H(\mathbf{A}^{-}\underline{\mathbf{b}}) + [(\mathbf{I} - \mathbf{A}^{-}\mathbf{A})\underline{\mathbf{x}}]^H[(\mathbf{I} - \mathbf{A}^{-}\mathbf{A})\underline{\mathbf{x}}] \\
&\quad + \underline{\mathbf{b}}^H(\mathbf{A}^{-})^H\underline{\mathbf{x}} - \underline{\mathbf{b}}^H(\mathbf{A}^{-}\mathbf{A}\mathbf{A}^{-})^H\underline{\mathbf{x}} \\
&= (\mathbf{A}^{-}\underline{\mathbf{b}})^H(\mathbf{A}^{-}\underline{\mathbf{b}}) + [(\mathbf{I} - \mathbf{A}^{-}\mathbf{A})\underline{\mathbf{x}}]^H[(\mathbf{I} - \mathbf{A}^{-}\mathbf{A})\underline{\mathbf{x}}] \\
&\quad + \underline{\mathbf{b}}^H(\mathbf{A}^{-})^H\underline{\mathbf{x}} - \underline{\mathbf{b}}^H(\mathbf{A}^{-})^H\underline{\mathbf{x}} \\
&[\mathbf{A}^{-}\underline{\mathbf{b}} + (\mathbf{I} - \mathbf{A}^{-}\mathbf{A})\underline{\mathbf{x}}]^H[\mathbf{A}^{-}\underline{\mathbf{b}} + (\mathbf{I} - \mathbf{A}^{-}\mathbf{A})\underline{\mathbf{x}}] \\
&= (\mathbf{A}^{-}\underline{\mathbf{b}})^H(\mathbf{A}^{-}\underline{\mathbf{b}}) + [(\mathbf{I} - \mathbf{A}^{-}\mathbf{A})\underline{\mathbf{x}}]^H[(\mathbf{I} - \mathbf{A}^{-}\mathbf{A})\underline{\mathbf{x}}] \quad (\text{all } \underline{\mathbf{x}} \text{ in } \mathbb{C}^N)
\end{aligned}$$

For $\mathbf{A}\mathbf{A}^{-}\underline{\mathbf{b}} = \mathbf{A}\underline{\mathbf{x}}$, we have $\mathbf{A}^{-}\underline{\mathbf{b}} = \mathbf{A}^{-}\mathbf{A}\underline{\mathbf{x}}$, so for equality (C - 1) to hold:

$$\begin{aligned}
&(\mathbf{A}^{-}\mathbf{A}\underline{\mathbf{x}} + \underline{\mathbf{x}} - \mathbf{A}^{-}\mathbf{A}\underline{\mathbf{x}})^H(\mathbf{A}^{-}\mathbf{A}\underline{\mathbf{x}} + \underline{\mathbf{x}} - \mathbf{A}^{-}\mathbf{A}\underline{\mathbf{x}}) = \underline{\mathbf{x}}^H\underline{\mathbf{x}} \\
&= (\mathbf{A}^{-}\underline{\mathbf{b}})^H(\mathbf{A}^{-}\underline{\mathbf{b}}) + (\underline{\mathbf{x}} - \mathbf{A}^{-}\underline{\mathbf{b}})^H(\underline{\mathbf{x}} - \mathbf{A}^{-}\underline{\mathbf{b}})
\end{aligned}$$

So, for all $\underline{\mathbf{x}}$ in \mathbb{C}^N such that equality (C - 1) holds, $\underline{\mathbf{x}}^H\underline{\mathbf{x}} > \underline{\mathbf{x}}_0^H\underline{\mathbf{x}}_0$ if $\underline{\mathbf{x}} \neq \underline{\mathbf{x}}_0$.

Q.E.D.

APPENDIX D

SIMULATION PROGRAM DESCRIPTION

The experimental results given in Chapter IV were produced using a simulation program developed for the OSU ECEN VAX 11/750. This program permits direction of arrival (DOA) determination experiments to be performed using computer-generated data. The program was developed in FORTRAN-77, for the VMS operating system (Version 4.5), and consists of an interactive routine and four batch routines. User interface functions are performed by the interactive routine, while the four batch routines perform various estimation functions associated with DOA estimation.

The five routines communicate information concerning DOA estimation experiments in progress by way of two index files and multiple direct and sequential files on disk. The MUSIC index file records information concerning individual experiments, such as number of antenna elements, arriving waves, and noise generators to simulate, the number of samples to process, sample interval, IRLS parameters, and the range of p values to support. The "spawn" index file records the status of the various batch routines used to perform the estimation. The other files (sequential and direct) used for communication record information about each experiment in progress, such as antenna element positions; wave strengths, angles of arrival and modulation envelope characteristics; noise generator strengths, distributions, and correlation coefficients; estimated directions of arrival; observation and noise sample vectors and

estimates; and computed values such as eigenvalues, eigenvectors, and covariance matrices.

Experiments recorded in the MUSIC index file can be linked in a fashion which permits one experiment to produce a number of related experiments. In the comment lines contained in the routines, an experiment which produces another experiment is called a "parent" experiment, and an experiment produced from another experiment is called a "descendant" experiment. This linking can encompass multiple levels of "descendancy", such that one experiment can produce a collection of descendant experiments, which in turn can each produce descendant experiments of their own, and so on. When an experiment gives rise to multiple levels of descendant experiments, the original experiment is called the "ancestor" experiment for each of the experiments in its descendancy chains.

The ancestor-descendant linkage mechanism is used to perform the iterative DOA revision described in Chapter III of this dissertation. This is accomplished as follows. Once the generalized eigenvalue problem is solved for a particular experiment, the operator is given the opportunity to select estimated directions of arrival from the experiment's DOA spectra. Once estimated DOAs are determined, the operator can initiate processing of new experiments which use the original experiment's basic parameters (number of antenna elements, etc.), but whose generalized eigenvalue problems are adjusted using the estimated DOAs just obtained. Each new experiment so initiated is a descendant of the original experiment, and is linked to it in the manner described in the previous paragraph. This process can be repeated for each of the descendant experiments if desired, and so the multiple levels of descendancy described above can be produced.

Routine DO_MUSIC

Program DO_MUSIC is designed to run in an interactive environment, and provides the operator interface to the L_p MUSIC algorithms. Lines 1 - 3117 contain comments and INCLUDEd declaration files. Setting the process name for coordination with the batch routines, obtaining logical unit numbers, opening the required index files, and other initialization are performed in lines 3118 - 3199. Lines 3200 - 3208 submit batch routines to perform sample generation, eigenproblem solution, or P matrix estimation as necessary. An index of active experiments is produced in lines 3209 - 3220. Lines 3221 - 3256 present the user with a menu of activities from which to select, and accept the operator's input selecting the activity to perform. Actions which can be performed are discussed in the following paragraphs.

If the operator selects generation of an index of active experiments, lines 3257 - 3264 set an indicator that an index is desired, and return to accept further operator input.

Lines 3265 - 3594 process a user command to determine DOA estimates for an experiment. User selection of the experiment for which to plot DOA spectra is handled in lines 3265 - 3322. Lines 3323 - 3350 locate the selected experiment's ancestor, if necessary. Lines 3351 - 3408 allocate arrays required for storage of the information necessary to plot DOA spectra, and obtain the necessary information from the appropriate sequential data files. A direct-access scratch file for DOA accumulation is opened in lines 3409 - 3416. Actual DOA estimation is accomplished in lines 3417 - 3547. Lines 3420 - 3438 accept the number of noise eigenvectors to use and the range of p values for which to plot spectra. Spectra plots are produced by subprogram MUSIC_PLOT_SPECT called at line 3443, and peak location is accomplished by

subprogram PICK_MUSIC_PEAK invoked at line 3447. Lines 3453 - 3528 initiate descendant experiments as necessary using the estimated DOAs. Finally, lines 3529 - 3594 process operator selections for the disposition of the experiment for which DOAs were estimated, discard space allocated for DOA information storage, and handle any file I/O errors that may have occurred during the DOA estimation process.

Printing of a selected experiment's output file is accomplished in lines 3595 - 3624, if selected by the operator.

Processing of the operator's selection to estimate a source covariance matrix is accomplished in lines 3625 - 3819. Lines 3625 - 3762 accept the operator's selection of which experiment to process, and accumulate the DOA estimates associated with that experiment, its parent, its parent's parent, and so on up to the ancestor experiment. Lines 3763 - 3799 accept the user's choice of the estimation type to be used, and spawn a batch job (MUSIC_PEST) to evaluate the source covariance matrix estimate. File I/O errors are handled in lines 3800 - 3819.

An experiment selected by the operator can be marked for deletion (from the index of experiments) in lines 3820 - 3895, while lines 3896 - 3998 actually delete the index entry and data files for an experiment that has been previously marked for deletion.

Display of results for an experiment is accomplished in lines 3999 - 4579. Lines 3999 - 4130 accept the operator's choice of what results are to be displayed, and the experiment for which to display the selected results. An array picture is generated in lines 4131 - 4199; lines 4200 - 4303 plot any combination of array output, source signal vector, source vector estimate, or noise vector samples; lines 4304 - 4425 plot selected DOA spectra; and lines 4426 - 4579

plot an operator-selected combination of array output samples, estimated source vector samples, noise vector samples, and DOA spectra.

Lines 4580 - 4626 permit operator re-initializing of an experiment that was accidentally marked for deletion.

If the user wishes to re-run an experiment that was previously entered (creating a new "trial"), this is accomplished in lines 4627 - 4778.

A new experiment is activated by the operator in lines 4779 - 5119. Lines 4779 - 5017 assign default values for the experiment. Default values can be taken from an already existing experiment, or may be set to "standard" default values, at the operator's selection. Operator adjustment of default values is accomplished in lines 5018 - 5039. The experimental parameters under operator control are antenna array geometry, arriving wave characteristics, and noise attributes.

The antenna array to be simulated can consist of any number of antenna elements arranged in any desired locations about a common origin. No constraints are placed on antenna locations; however, all simulated antenna elements are identical and omnidirectional. Once the receiving array has been described, the operator can specify any number of arriving waves. Wave characteristics under operator control are angle of arrival (azimuth only), signal strength, carrier wavelength, and modulation characteristics. The program supports both amplitude and phase modulation of the received signals, and permits operator selection of modulation envelope types, lengths, duty cycles, and relative phases.

The final simulation aspect under operator control is sample noise characteristics. Gaussian, uniform, and fixed noise generators of operator-selected intensities are available, the outputs from any number of which may be applied to any combination of antenna elements. The noise generators may be

used on either a continuous or intermittent basis, permitting simulation of noise or interference bursts impinging on the array. Two separate noise models are maintained by the program. One model is used in the estimation routines (e.g. DOA determination), and the other is used in actual sample generation. The two models may be identical or different, at the operator's option.

After operator adjustment of default experiment values, the new experiment is actually initiated in lines 5040 - 5107. Lines 5108 - 5119 handle file I/O errors that may occur during experiment generation.

An operator command to exit to the operating system is processed in lines 5120 - 5128, and lines 5129 - 5135 handle an invalid operator selection.

Lines 5136 - 5164 are used at various points in DO_MUSIC to update an experiment's index file entry.

File I/O errors not handled elsewhere are processed in lines 5165 - 5260.

Several subprograms are included in the DO_MUSIC source file to accomplish various tasks needed by the operator interface. Function P_INDEX finds and displays records for selected experiments in the experiment index file; function NEXT_REC is used by P_INDEX to locate the next selected index record; batch jobs to be initiated are spawned by subroutine SUBMIT_JOB; and PRINT_EXPM initiates printing of output files for selected experiments.

Routine SAMPL_MUSIC

Routine SAMPL_MUSIC, which was designed to run as a batch program, performs the sample generation function for experiments in progress. Lines 1 - 610 perform required initialization functions. Line 483 selects an experiment for which sample generation is required. Input of experiment definition data is accomplished in lines 611 - 660. Lines 661 - 743 compute the observation and noise vector samples, write the samples to the appropriate data files, and print

the problem definition information to the experiment's output file. Lines 744 - 812 compute covariance matrices and the minimum eigenvalue estimate, perform QR decompositions of the array output and noise sample matrices, write the results of the QR decompositions to the appropriate data file, and print the covariance matrices to the experiment's output file. Lines 813 - 863 update the experiment's index file entry and initiate the eigenvector solution routine if necessary. Lines 864 - 875 terminate processing if no more experiments require sample generation, and lines 876 - 912 process any file I/O errors.

Routine EIG_MUSIC

Routine EIG_MUSIC performs the eigenvector decomposition required by the L_p MUSIC approach. It is designed to run as a batch procedure. Lines 1 - 464 perform required initialization. Selection of an experiment for which to perform eigenvector decomposition, and input of the data necessary to accomplish the decomposition, is performed in lines 465 - 565. Lines 566 - 597 solve the generalized eigenvalue problem, write the results to appropriate data files, perform a check of the results, and print various computed quantities to the experiment's output file. Lines 598 - 643 update the experiment's index file entry, and terminate processing if no more experiments require eigenvector decomposition. Finally, lines 644 - 708 process any file I/O errors which may have occurred.

Routine FIN_MUSIC

This routine, designed to operate as a batch routine, obtains estimates of source signal vector samples using observation vector samples and DOA estimates determined in DO_MUSIC. Initialization is performed in lines 1 - 482. Lines 483 - 734 select the experiment to be processed and obtain the data

necessary for the estimation process from the appropriate data files. Lines 735 - 746 evaluate the sharpness measures for the DOAs being used, and estimate the antenna array phase shift matrix from the estimated DOAs. Lines 747 - 788 estimate the source signal vector samples and source covariance matrix using the estimation type selected by the operator (in routine DO_MUSIC). Updating of the observation covariance matrix for iterative DOA revision is accomplished in lines 789 - 827. Finally, results are printed to the experiment's output file and written to the appropriate data files in lines 828 - 883. Lines 885 - 938 initiate the eigenvector decomposition routine, update the experiment's index file entry, and terminate processing if no more experiments require sample estimation. File I/O errors are handled in lines 939 - 985.

Routine MUSIC_PEST

This routine, which operates as a batch program, estimates the source covariance matrix for an experiment and evaluates the performance of the estimation process. Performance is evaluated by comparing the source covariance matrix estimate with the actual sample source covariance matrix. The actual sample source covariance matrix is computed by approximating the definition:

$$\mathbf{P} \approx \frac{1}{S} \sum_{s=1}^S \mathbf{f}_s \mathbf{f}_s^H$$

where:

S is the number of samples, and

\mathbf{f}_s is the s^{th} sample of the source signal vector.

The performance comparison is made using an error measure produced from the Frobenius norm of the difference between the matrix estimate and the actual sample source covariance matrix. The error measure used is:

$$\varepsilon = \frac{\sum_j \sum_k |\hat{p}_{j,k} - p_{j,k}|^2}{\sum_j \sum_k |p_{j,k}|^2}$$

where:

$\hat{p}_{j,k}$ is the (j,k) th element of the covariance matrix estimate, and $p_{j,k}$ is the (j,k) th element of the actual sample source covariance matrix.

Lines 1 - 467 perform initialization, and lines 468 - 494 obtain an input file if any is available. If no input files are available, indicating that no experiments require performance evaluation, line 487 terminates execution. Lines 495 - 677 read the information required for the estimation process from the appropriate data files. Actual estimation and performance evaluation are conducted in lines 678 - 746. Results are written to the selected experiment's output file in lines 747 - 762. Finally, lines 764 - 770 delete the processed input file, and lines 771 - 824 process any errors that may have occurred during the estimation process.

Other Routines

Various subprograms are used by the above-described routines to accomplish the estimation functions required in the L_p MUSIC process. The purposes of these subprograms are listed below.

- BRACKET_DOA: Evaluates DOA peak sharpness measures.
- C16_VACHOL: Performs Cholesky decomposition of updated \mathbf{R}_x .
- C8_NOISE: Generates noise vector samples.
- DCV_IRLS: Performs IRLS estimation of a COMPLEX*16 vector.
- RESIDUALS: Used by DCV_IRLS to compute estimation residuals.

DC_MPPSINV:	Evaluates the Moore-Penrose psuedo-inverse of a COMPLEX*16 matrix using singular value decomposition.
DCSV_VSIGM1:	Used by DC_MPPSINV to evaluate a matrix product.
DUMP_VA:	Discards all storage allocated by the Virtual Array system (copyright 1987 by Dr. Dwight Day).
EIG_FROM_SV:	Converts results of generalized singular value decomposition to generalized eigenvalue problem solution.
ENDSPAWN_IF:	Terminates program execution if all criteria for doing so have been met.
EXIT_VA:	Processes status conditions returned by Virtual Array routines (copyright 1987 by Dr. Dwight Day).
FREE_VA:	Releases specific portions of memory allocated by the Virtual Array system (copyright 1987 by Dr. Dwight Day).
GET_MUSIC_DEF:	Obtains experiment definition values from the user.
GET_W_HALF:	Evaluates the square root of a diagonal weight matrix.
MAKE_NOISE:	Computes a COMPLEX*16 noise sample vector.
MODULATED:	Returns the complex amplitude of the modulation envelope of an arriving wave at a specified sample instant.
MOD_VAL:	Used by MODULATED to obtain amplitude.
MOD_PHASE:	Used by MODULATED to obtain phase.
MUSIC_INIT:	Computes complex amplitudes of arriving waves and initializes noise counters.
MUSIC_L2P:	Computes an L_2 estimate of the \mathbf{P} matrix.
MUSIC_LPP_EACH:	Computes L_p estimates of signal vector samples.
MUSIC_LPP_SPEIS:	Computes L_p estimate of \mathbf{P} using sum-of-powers method.

MUSIC_LPW_YAR:
Computes square root of diagonal weight matrix for sum-of-powers method.

YAR_ESTS: Used by MUSIC_LPW_YAR to compute signal vector estimates.

YAR_RESIDS: Used by MUSIC_LPW_YAR to compute estimation residuals.

YAR_WEIGHTS: Used by MUSIC_LPW_YAR to compute weights.

MUSIC_PLOT_SPECT:
Plots DOA spectra for various p values.

VALUE_AT: Used by MUSIC_PLOT_SPECT to return a selected vector element.

MUSIC_POWER: Evaluates a DOA spectrum value.

MPWR_FL5, MPWR_DPDTRM, MPWR_FL1, MPWR_DPDSUM, MPWR_PWRTRM, MPWR_FL3, MPWR_PWRSUM:
Used by MUSIC_POWER to evaluate spectrum value and handle floating-point exceptions.

MUSIC_SPECTRUMX:
Evaluates abscissa values for a spectrum plot.

MUSIC_SPECTRUMY:
Evaluates ordinate values for a spectrum plot.

NOISE_COMBINE: Combines noise generator outputs to form antenna noise vector sample.

OBSVEC: Computes observation vector sample.

OPEN_VA: Dynamically allocates storage for FORTRAN programs; part of the Virtual Array system (copyright 1987 by Dr. Dwight Day).

PICK_MUSIC_PEAK: Determines the location of peaks in DOA spectra.

READ_ANTENNAE, READ_LP, READ_MMISC, READ_NOISE, READ_WAVES:
Obtain experiment definition values from the user.

PR_NTTYPE: Used by READ_NOISE to prompt for noise type.

PR_MTYPE: Used by READ_WAVES to prompt for modulation type.

SIGMAT_ESTM: Form estimate of array phase shift matrix.

SIGMAT_FORM: Form actual phase shift matrix.

SIGVAL: Compute phase shift vector for an incoming wave.

SIGVEC: Compute source signal vector at a selected sample instant.

SPAWN_IF: Initiates a selected batch job if it is not already active.

SPECT_PEAK: Determine the location of a peak in a DOA spectrum.

SQ_VLGSVD: Compute generalized singular value decomposition using Van Loan's method.

VA_DC_QR: Compute QR decomposition of a COMPLEX*16 matrix.

VA_DC_SQSVD: Compute singular value decomposition of a square COMPLEX*16 matrix.

VA_DC_TRINV: Compute the inverse of an upper triangular COMPLEX*16 matrix.

APPENDIX E

SIMULATION PROGRAM LISTING

This appendix provides computer listings of the routines which comprise the simulation program described in Appendix D. A scaled-down version of the simulation program listed herein is in development for use on personal computers.

```

c
c*****
c
c      This program segment defines various values for use in
c      accessing a MUSIC experiment index file. A MUSIC experiment index file
c      has the following characteristics (note that the description below can
c      be inserted into an "open" statement if the 1st columns are cleared):
c
c      2      form='UNFORMATTED',recordtype='FIXED',recl=64,
c      3      organization='INDEXED',access='KEYED',
c      4      key=(1:4: INTEGER,5:5: CHARACTER,6:9: INTEGER),
c      5      dispose='KEEP',SHARED)
c
c      (The keyword SHARED may be replaced by READONLY). A read / write from
c      / to this file takes the following form:
c
c      read (...)
c      1      mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
c      2      mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
c      3      mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
c      4      mi_tol, mi_which, mi_ofile, mi_anaises, mi_irislim,
c      5      mi_zeigs, mi_doas, mi_estype, mi_parp
c
c      The fields in a MUSIC experiment index file record are defined
c      as follows:
c
c      Byte Positions      Description
c      -----
c      1 - 4      mi_pkey: Primary key value, INTEGER*4 (unique record
c                  number).
c      5 - 5      mi_stat: Experiment status, CHARACTER*1:
c                  '-' = new experiment, no files created
c                  (or input file creation failed).
c                  ' ' = new experiment, input file (but
c                  no sample files) created.
c                  'I' = initialized, sample files
c                  created (ready for eigenvalue
c                  problem computations).
c                  'E' = eigenvalue problem solved (ready
c                  for DOA selection).
c                  'D' = DOA selections made (ready for Lp
c                  estimation process) (Note that
c                  this is the same status for
c                  "descendant" experiments as ' '
c                  is for "ancestor" experiments).
c                  'R' = experiment complete; results
c                  displayed (at least once).
c                  'X' = experiment failed; computational
c                  error encountered.
c      6 - 9      mi_park: INTEGER*4 "mi_pkey" of the "parent"
c                  experiment (the experiment which
c                  produced the residuals which are used
c                  as observations in this record's
c                  experiment) (-1 for an "original" or
c                  "ancestor" experiment (one with no
c                  "parent")).
c      10 - 11     mi_ierr: INTEGER*2 number of file read failures
c                  encountered by the batch sample
c                  generator job (each failure causes

```

```

      status to change from ' ' to '-').
c      c      12 - 13    mi_eerr:  INTEGER*2 number of file read failures
c      c      c      encountered by the batch eigenvalue
c      c      c      problem solver job (each failure causes
c      c      c      status to change from 'I' to '-').
c      c      14 - 15    mi_serr:  INTEGER*2 number of file read failures
c      c      c      encountered by the batch Lp estimation
c      c      c      job (each failure causes status to
c      c      c      change from 'D' to 'E').
c      c      16 - 34    mi_dtim:  CHARACTER*19 date / time of experiment.
c      c      35 - 39    mi_ants:  INTEGER*4 number of antenna elements in the
c      c      c      experiment.
c      c      39 - 42    mi_waves:  INTEGER*4 number of arriving waves in the
c      c      c      experiment.
c      c      43 - 46    mi_noises:  INTEGER*4 number of noise sources used in
c      c      c      the ESTIMATION noise model for the
c      c      c      experiment.
c      c      47 - 50    mi_smp:   INTEGER*4 number of samples in the experiment.
c      c      51 - 58    mi_sint:  REAL*8 sample interval for the experiment.
c      c      59 - 62    mi_begp:  REAL*4 starting value of p for Lp estimation.
c      c      63 - 66    mi_endp:  REAL*4 ending value of p for Lp estimation
c      c      c      (< mi_begp for no Lp estimation).
c      c      67 - 70    mi_delp:  REAL*4 p increment value for Lp estimation.
c      c      71 - 78    mi_peps:  REAL*8 minimum residual value for Lp
c      c      c      estimation.
c      c      79 - 86    mi_norm:  REAL*8 value to which to normalize Lp
c      c      c      residuals (0 for unnormalized).
c      c      87 - 94    mi_wlen:  REAL*8 carrier wavelength for the experiment.
c      c      95 - 102   mi_tol:  REAL*8 value to be taken for zero
c      c      c      ("tolerance").
c      c      103 - 106   mi_which:  INTEGER*4 selector which determines whether
c      c      c      estimators use actual (mi_which = 0) or
c      c      c      model (mi_which > 0) noise samples.
c      c      107 - 186   mi_ofile:  CHARACTER*80 name of the experiment's output
c      c      c      file.
c      c      187 - 190   mi_anoses:  INTEGER*4 number of noise sources used in
c      c      c      sample GENERATION.
c      c      191 - 194   mi_irlslim:  INTEGER*4 IRLS iteration limit.
c      c      195 - 198   mi_zeigs:  INTEGER*4 number of "noise" eigenvectors
c      c      c      used (in the "parent" experiment) to
c      c      c      determine the DOAs which give rise to
c      c      c      the residuals which serve as
c      c      c      observations in this record's
c      c      c      experiment (0 for "ancestor"
c      c      c      experiment).
c      c      199 - 202   mi_doas:  INTEGER*4 number of DOAs determined (in the
c      c      c      "parent" experiment), which are used to
c      c      c      produce the residuals which serve as
c      c      c      observations in this record's
c      c      c      experiment (0 for "ancestor"
c      c      c      experiment).
c      c      203 - 206   mi_estype:  INTEGER*4 value specifying the type of
c      c      c      estimation used to produce the
c      c      c      residuals which serve as observations
c      c      c      in this record's experiment (-1 for
c      c      c      "ancestor" experiment):
c      c      c      1: Enumeration Method
c      c      c      2: Sum-of-Powers Method
c      c      c      3: Compensated Enumeration
c      c      c      4: Sum-of-Powers Loss Method

```



```

c                                5: Compensated Sum-of-Powers
c                                6: Comp'd Sum-of-Powers Loss
c      207 - 210:   mi_parp: REAL*4 value of p used in the "parent"
c                                experiment to produce the residuals
c                                which serve as observations for this
c                                record's experiment (0.00 for
c                                "ancestor" experiment).
c
integer*2 mi_ierr, mi_eerr, mi_serr
integer*4 mi_pkey, mi_park, mi_ants, mi_waves, mi_noises, mi_smp
integer*4 mi_which, mi_anoises, mi_irlslim, mi_zeigs, mi_doas
integer*4 mi_estype
real*4 mi_begp, mi_endp, mi_delp, mi_parp
real*8 mi_sint, mi_peps, mi_norm, mi_wlen, mi_tol
character*1 mi_stat
character*19 mi_dtim
character*80 mi_ofile
c
c*****
c

```

```

c
c      This program segment contains the declarations used by the
c      MUSIC / Lp main program.
c
intrinsic date, time
intrinsic index, char
external init_va
external lib$get_lun, lib$stop, lib$delete_file, lib$getjpi
external lib$find_file, lib$find_file_end
external cap, mreaddoas
external open_va, exit_va, dump_va
external init_music_def, get_music_def, avlbi_prikey
external writearr, readarr, skiparr
external read_antennae, read_waves, read_noise, read_mmisc, read_lp
external dr_colfill
external lib$wait
external type_eigs
external music_plot_inps, music_picture, music_plot_spect
external music_plot_ests, music_plot_trial, collect_doas
external pick_music_peak, prange
c
logical writearr, readarr, skiparr
logical music_plot_spect, music_plot_ests, music_plot_trial
logical collect_doas
integer*4 lib$get_lun, avlbi_prikey, lib$getjpi, type_eigs
integer*4 lib$find_file, lib$find_file_end
character*1 cap, pick_music_peak
c
logical disp_index, havent_irec, pinfo_shown, oinst_shown, dummy
logical ancestor
integer*2 tempi
integer*4 err_return      ! Statement # for return from data I/O error.
integer*4 old_ants, old_waves, old_noises
integer*4 old_anoises, old_which, old_smp
integer*4 vstatus, status, pname_len
integer*4 data_unit, index_unit, read_unit1, read_unit2, spawn_unit
integer*4 data_unit2
integer*4 ants_lin( 1 ), ants_sqr( 2 ), ants_by_2( 2 )
integer*4 waves_lin( 1 ), waves_by_9( 2 )
integer*4 waves_by_2( 2 )
integer*4 nois_by_2( 2 ), nois_by_3( 2 )
integer*4 ants_by_nois( 2 )
integer*4 smp_by_ants( 2 ), ants_by_smp( 2 )
integer*4 wavs_by_smp( 2 ), doas_lin( 1 )
integer*4 anois_by_2( 2 ), anois_by_3( 2 ), ants_by_anois( 2 )
integer*4 ant_add
integer*4 wave_add, mtype_add
integer*4 ndef_add, npars_add, correl_add
integer*4 obsmat_add, noisemat_add
integer*4 sigvm_add
integer*4 clinmat_add
integer*4 andef_add, anpars_add, acorrel_add
integer*4 eig_add, eigv_add
integer*4 pvm_add
integer*4 resm_add
integer*4 yvm_add, yresm_add, doa_add, exnm_add
integer*4 i, count, fskip, iskip, num_zelgs, num_doas
integer*4 num_acc, num_exps
integer*4 vsel, elsel, selexp, selpar, itemp, estype/ 3 /
real*4 r_answer, p

```

```
character*1 psel, tch
character*1 uparrow( 3 ) / 27, 'L', 'A' /, spaces( 70 ) / 70* ' ' /
character*3 chsel
character*5 ptstat( 8 ) / 'IEDR', 'IEDR/', 'IEDR/', 'IEDR/',
1 'IEDR/', 'IEDR/', 'ER////', 'ER////' /
character*10 answer
character*13 ptypes( 8 ) / 'array picture', 'signal-in-spc',
1 'clean observ.', 'noisy observ.',
2 'observ. noise', 'all input val',
3 'dir. of arriv', 'all trial val' /,
4 p_etype( 6 ) / 'EJ/P/N/L/S/C', 'E/CPJ/N/L/S/C',
5 'E/P/INJ/L/S/C', 'E/P/N/LLJ/S/C',
6 'E/P/N/L/CSJ/C', 'E/P/N/L/S/CCJ' /
character*15 procname
character*25 work_file
character*80 msg, ti_ofile
```

c

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```

0001      program do_music      ! MAIN program for MUSIC / Lp exercises.
0002      c
0003      c*****
0004      c
0005      c      This comment section describes the Virtual Array System vectors
0006      c      and matrices used by the MUSIC / Lp algorithm. The name of each vector
0007      c      (matrix) address is the "name" of the vector (matrix) with the
0008      c      characters '_add' appended to it. Thus, the vector ant starts at
0009      c      memory location ant_add, and so on.
0010      c      The Virtual Array System has been copyrighted by Dwight Day in
0011      c      1987.
0012      c
0013      c*****
0014      c
0015      c      ---- ant:  ants x 2 REAL*8 matrix whose first column is the vector
0016      c      of antenna ranges from the origin, and whose
0017      c      second column is the vector of antenna angles
0018      c      from the abscissa (RADIANS).
0019      c      ---- wave:  waves x 9 REAL*8 matrix whose columns are as follows:
0020      c      1) vector of source signal strengths in db.
0021      c      2) vector of source signal carrier starting
0022      c      phases (RADIANS).
0023      c      3) vector of actual angles of arrival of the
0024      c      incoming waves.
0025      c      4) vector of incoming wave modulation cycle
0026      c      lengths, in carrier cycles.
0027      c      5) vector of incoming wave modulation duty
0028      c      cycles, as a fraction from 0 to
0029      c      1.
0030      c      6) vector of incoming wave modulation
0031      c      (envelope) start phases
0032      c      (carrier cycles).
0033      c      7) vector of incoming wave amplitude modulation
0034      c      ratios (maximum amplitude /
0035      c      minimum amplitude).
0036      c      8) vector of phase modulation total phase
0037      c      shifts for the incoming waves
0038      c      (RADIANS).
0039      c      9) vector of carrier wavelengths of the
0040      c      incoming waves.
0041      c      ---- ccomp:  waves-element COMPLEX*16 vector of source complex
0042      c      amplitudes (in "rectangular" form).
0043      c      ---- mtype:  waves x 2 INTEGER*4 matrix whose first column is the
0044      c      vector of amplitude modulation types for
0045      c      incoming waves, and whose second column is the
0046      c      vector of phase modulation types for the
0047      c      incoming waves.
0048      c      ---- ndef:  noises x 2 INTEGER*4 matrix whose first column is the
0049      c      vector of noise types, and whose second column
0050      c      is the vector of noise counters (for
0051      c      intermittent noise). This matrix is used for
0052      c      the ESTIMATION noise model.
0053      c      ---- andef:  anoises x 2 INTEGER*4 matrix whose first column is the
0054      c      vector of noise types, and whose second column
0055      c      is the vector of noise counters (for
0056      c      intermittent noise). This matrix is used for
0057      c      ACTUAL SAMPLE GENERATION.

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0058 c ----- npars: noises x 3 REAL*4 matrix of noise parameters (mean,
0059 c          std. dev., ave. rep. rate). This matrix is
0060 c          used for the ESTIMATION noise model.
0061 c ----- anpars: anoises x 3 REAL*4 matrix of noise parameters (mean,
0062 c          std. dev., ave. rep. rate). This matrix is
0063 c          used for ACTUAL SAMPLE GENERATION.
0064 c ----- correl: ants x noises REAL*8 matrix of noise multipliers for
0065 c          the antenna array. This matrix is used for the
0066 c          ESTIMATION noise model.
0067 c ----- acorrel: ants x anoises REAL*8 matrix of noise multipliers for
0068 c          the antenna array. This matrix is used for
0069 c          ACTUAL SAMPLE GENERATION.
0070 c ----- obsmat: smp x ants COMPLEX*16 matrix whose ROWS are the
0071 c          samples of the antenna array output (including
0072 c          noise).
0073 c ----- noisemat: smp x ants COMPLEX*16 matrix whose ROWS are the
0074 c          noise samples used in the ESTIMATION process.
0075 c ----- rb: ants x ants noise covariance matrix, normalized such
0076 c          that its trace is ants. This matrix is
0077 c          produced from noisemat; that is, the noise to
0078 c          be used for ESTIMATION.
0079 c ----- obs: ants-element COMPLEX*16 vector sample of antenna array
0080 c          output (including noise).
0081 c ----- noise: ants-element COMPLEX*16 vector noise sample.
0082 c ----- robs: ants x ants COMPLEX*16 observation (including noise)
0083 c          covariance matrix.
0084 c ----- rnoise: ants x ants COMPLEX*16 noise covariance matrix (not
0085 c          trace normalized). This is the covariance
0086 c          matrix of the noise produced from the
0087 c          ESTIMATION noise model.
0088 c ----- arnoise: ants x ants COMPLEX*16 noise covariance matrix (not
0089 c          trace normalized). This is the covariance
0090 c          matrix of the noise actually used in GENERATING
0091 c          the antenna array output samples.
0092 c ----- ants_di: ants x ants COMPLEX*16 work matrix.
0093 c ----- sigvm: waves x smp COMPLEX*16 matrix whose COLUMNS are the
0094 c          actual or estimated signal-in-space (source
0095 c          amplitude & phase) vector samples.
0096 c ----- clinmat: ants x smp COMPLEX*16 matrix whose COLUMNS are the
0097 c          samples of the antenna array output (without
0098 c          noise).
0099 c ----- sspc: waves-element COMPLEX*16 signal-in-space (source
0100 c          amplitude and phase) vector sample.
0101 c ----- sgmt: ants x waves COMPLEX*16 signal coefficient ("A") matrix
0102 c          (actual).
0103 c ----- clin: ants-element COMPLEX*16 antenna array output sample
0104 c          (without noise).
0105 c ----- obst: ants x ants COMPLEX*16 upper-triangular result of QR
0106 c          decomposition of "obsmat".
0107 c ----- noiset: ants x ants COMPLEX*16 upper-triangular result of QR
0108 c          decomposition of "noisemat".
0109 c ----- eig: ants-element REAL*8 vector of generalized eigenvalues.
0110 c ----- eigv: ants x ants COMPLEX*16 matrix whose COLUMNS are the
0111 c          generalized eigenvectors corresponding to "eig"
0112 c          entries.
0113 c ----- amat: ants x waves COMPLEX*16 estimate of the signal
0114 c          coefficient ("A") matrix.

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0115 c ----- pmat: waves x waves COMPLEX*16 source covariance matrix
0116 c (actual).
0117 c ----- pl2: waves x waves COMPLEX*16 least-squares estimate of the
0118 c source covariance matrix.
0119 c ----- plp: waves x waves COMPLEX*16 "Lp" estimate (using L2 "A") of
0120 c the source covariance matrix.
0121 c ----- yarw: ants-element REAL*8 vector of final weights produced
0122 c by the "Yarlagadda" P-matrix IRLS algorithm.
0123 c
0124 c*****
0125 c
0126 c implicit none
0127 c external p_index, submit_job, print_expm
0128 c
0129 c integer*4 p_index
0130 c
0131 c include '6000*RT:[CHUCK.RESEARCH.FORTDIRJVTYPES.TXT'
0132 c include '6000*RT:[CHUCK.RESEARCH.FORTDIRJVL_MUSIC_DECL.FOR'
0221 c include '6000*RT:[CHUCK.RESEARCH.FORTDIRJMUSIC_MEASURES.FOR'
0310 c include '6000*RT:[CHUCK.RESEARCH.FORTDIRJMUSIC_INDEX_DEF.TXT'
0451 c include '6000*RT:[CHUCK.RESEARCH.FORTDIRJSPAWNED_DEF.TXT'
0494 c include 'SYS$LIBRARY:FORIOSDEF'
0580 c include '(system_symbols)'
0698 c include '($SYSSRVNAM)'
3118 c
3119 c Get current process name, and change process name to
3120 c "Chuck MUSIC Drv".
3121 c
3122 c status = lib$getjpi( JPI$_PRCNAM, , , , procname, pname_len )
3123 c if (.not. status) call lib$stop( %val( status ) )
3124 c status = sys$setprn( 'Chuck MUSIC Drv' )
3125 c if (.not. status) call lib$stop( %val( status ) )
3126 c
3127 c Get logical unit numbers and indicate that the program "just
3128 c woke up".
3129 c
3130 c status = lib$get_lun( data_unit )
3131 c if (.not. status) call lib$stop( %val( status ) )
3132 c status = lib$get_lun( data_unit2 )
3133 c if (.not. status) call lib$stop( %val( status ) )
3134 c status = lib$get_lun( index_unit )
3135 c if (.not. status) call lib$stop( %val( status ) )
3136 c status = lib$get_lun( read_unit1 )
3137 c if (.not. status) call lib$stop( %val( status ) )
3138 c status = lib$get_lun( read_unit2 )
3139 c if (.not. status) call lib$stop( %val( status ) )
3140 c status = lib$get_lun( spawn_unit )
3141 c if (.not. status) call lib$stop( %val( status ) )
3142 c
3143 c pinfo_shown = .false.
3144 c oinst_shown = .false.
3145 c disp_index = .true.
3146 c
3147 c Open the MUSIC index file.
3148 c
3149 c open (unit=index_unit,file='6000*RT:[CHUCK.PARAMS]MUSIC_INDEX.DAT',
3150 c 1 form='UNFORMATTED',recordtype='FIXED',recl=64,

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3151      2      organization='INDEXED',access='KEYED',status='OLD',
3152      3      key=(1:4:INTEGER,5:5:CHARACTER,6:9:INTEGER),
3153      4      dispose='KEEP',SHARED,err=30000)
3154      go to 30001
3155      c
3156      c      Error opening the MUSIC index file; create a new one.
3157      c
3158      30000      continue
3159      open (unit=index_unit,file='6000*RT:[CHUCK.PARAMS]MUSIC_INDEX.DAT',
3160      1      form='UNFORMATTED',recordtype='FIXED',recl=64,
3161      2      organization='INDEXED',access='KEYED',status='NEW',
3162      3      key=(1:4:INTEGER,5:5:CHARACTER,6:9:INTEGER),
3163      4      dispose='KEEP',SHARED)
3164      c
3165      c      MUSIC index file open, open again on a separate LUNs for
3166      c      display routines.
3167      c
3168      30001      continue
3169      open (unit=read_unit1,file='6000*RT:[CHUCK.PARAMS]MUSIC_INDEX.DAT',
3170      1      form='UNFORMATTED',recordtype='FIXED',recl=64,
3171      2      organization='INDEXED',access='KEYED',status='OLD',
3172      3      key=(1:4:INTEGER,5:5:CHARACTER,6:9:INTEGER),
3173      4      dispose='KEEP',SHARED)
3174      open (unit=read_unit2,file='6000*RT:[CHUCK.PARAMS]MUSIC_INDEX.DAT',
3175      1      form='UNFORMATTED',recordtype='FIXED',recl=64,
3176      2      organization='INDEXED',access='KEYED',status='OLD',
3177      3      key=(1:4:INTEGER,5:5:CHARACTER,6:9:INTEGER),
3178      4      dispose='KEEP',SHARED)
3179      c
3180      c      Open the MUSIC spawned processes index file.
3181      c
3182      open (unit=spawn_unit,file='6000*RT:[CHUCK.PARAMS]SPAWNED.IDX',
3183      1      form='UNFORMATTED',recordtype='FIXED',recl=26,
3184      2      organization='INDEXED',access='KEYED',status='OLD',
3185      3      key=(1:4:INTEGER,5:8:INTEGER),dispose='KEEP',SHARED,err=30100)
3186      go to 20000
3187      c
3188      c      Error opening spawned-processes index file. Give up.
3189      c
3190      30100      continue
3191      type *
3192      type *, 'Couldn't open file 6000*RT:[CHUCK.PARAMS]SPAWNED.IDX.'
3193      type *, 'Have you run the right version of MAKE_SPAWN_IDX?'
3194      type *
3195      status = sys$setprn( procname( 1 : pname_len ) )
3196      if (.not. status) call lib$stop( %val( status ) )
3197      stop
3198      c
3199      20000      continue      ! Branch for not first time through program.
3200      c
3201      c      Submit any required batch jobs.
3202      c
3203      if ( p_index( index_unit, read_unit1, read_unit2, ' ', .false. )
3204      1      .ge. 0 ) call submit_job( spawn_unit, 1 )
3205      if ( p_index( index_unit, read_unit1, read_unit2, 'I', .false. )
3206      1      .ge. 0 ) call submit_job( spawn_unit, 2 )
3207      if ( p_index( index_unit, read_unit1, read_unit2, 'D', .false. )

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3208      1                .ge. 0 ) call submit_job( spawn_unit, 3 )
3209      c
3210      c                Display the MUSIC index file if selected.
3211      c
3212      if ( disp_index ) then
3213          disp_index = .false.
3214          mi_pkey = p_index( index_unit, read_unit1, read_unit2, '*', .true. )
3215          if ( mi_pkey .eq. -1 ) then
3216              type *
3217              type *, 'No active MUSIC experiments.'
3218              type *
3219          end if
3220      end if
3221      c
3222      c                Allow user to choose the activity he wishes to perform.
3223      c
3224      type *
3225      type *, 'Please tell me what you wish to do:'
3226      type *, '  N (or n) to initiate a new experiment.'
3227      if ( p_index( index_unit, read_unit1, read_unit2, '-X', .false. )
3228          .ge. 0 ) type *, '  I (or i) to re-start aborted or failed ',
3229          'experiment.'
3230      if ( p_index( index_unit, read_unit1, read_unit2, 'ER', .false. )
3231          .ge. 0 ) then
3232          type *, '  D (or d) to determine DOAs for active experiment.'
3233          type *, '  P (or p) to print output file for completed experiment.'
3234      end if
3235      if ( p_index( index_unit, read_unit1, read_unit2, 'IEDR', .false. )
3236          .ge. 0 ) type *, '  C (or c) to evaluate performance for ',
3237          'completed experiment.'
3238      if ( p_index( index_unit, read_unit1, read_unit2, 'IEDR', .false. )
3239          .ge. 0 ) type *, '  L (or l) to display data for active ',
3240          'experiment.'
3241      if ( p_index( index_unit, read_unit1, read_unit2, '-ER', .false. )
3242          .ge. 0 ) type *, '  M (or m) to mark active experiment for ',
3243          'deletion.'
3244      if ( p_index( index_unit, read_unit1, read_unit2, '-RX', .false. )
3245          .ge. 0 ) type *, '  K (or k) to delete completed or failed ',
3246          'experiment.'
3247      if ( p_index( index_unit, read_unit1, read_unit2, 'IER', .false. )
3248          .ge. 0 ) type *, '  R (or r) to re-run an initialized ',
3249          'experiment (new "trial").'
3250      type *, '  S (or s) to display current status of all experiments.'
3251      type *, '  X (or x) to exit to the operating system.'
3252      c
3253      type *
3254      type 90000, ' What does the master command? '
3255      accept 90010, answer
3256      psel = cap( answer( 1 : 1 ) )
3257      c
3258      c!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!
3259      c
3260      c                Status of active experiments selected.
3261      c
3262      if      ( psel .eq. 'S' ) then
3263          disp_index = .true.
3264          go to 20000

```



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3265  c                                     !
3266  c!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!
3267  c                                     !
3268  c           Determine DOAs of active experiments selected.                !
3269  c                                     !
3270                                     else if ( psel .eq. 'D' ) then
3271  type *
3272  type *, '                               DOA SPECTRUM PEAK SELECTIONS'
3273  type *
3274  if ( .not. pinfo_shown ) then
3275  60420  continue                               ! Print plot info.
3276  pinfo_shown = .true.
3277  type *, 'During plotting, the computer types an appropriate ',
3278  1      type *, 'then waits for a <CR> before proceeding with the plot. ',
3279  1      type *, 'finished, the computer waits for another <CR> before ',
3280  1      type *, 'program step.'
3281  1      type *, 'continuing with the next'
3282  type *
3283  type *
3284  type *
3285  end if
3286  type *, 'Enter experiment # for which to determine DOAs,'
3287  type *, '      <CR> to see a list of available experiments,'
3288  type 90000, '                               ? for information about plotting: '
3289  accept 90010, answer
3290  if ( index( answer, '?' ) .gt. 0 ) go to 60420
3291  read (unit=answer,fmt=*,err=60430) mi_pkey
3292  if ( mi_pkey .ge. 0 ) go to 60410
3293  c
3294  60430  continue                               ! Error in accepted #.
3295  mi_pkey
3296  1      = p_index( index_unit, read_unit1, read_unit2, 'ER', .true. )
3297  if ( mi_pkey .eq. -1 ) then
3298  60000  continue                               ! None available.
3299  type *
3300  type *, char( 7 ), 'Can't plot DOA spectrum.'
3301  type *
3302  go to 20000
3303  end if
3304  if ( mi_pkey .lt. 0 ) go to 20000           ! Operator cancelled.
3305  c
3306  60410  continue
3307  read (unit=index_unit,keyeq=mi_pkey,keyid=0,iostat=status)
3308  1      mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
3309  2      mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
3310  3      mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
3311  4      mi_tol, mi_which, mi_ofile, mi_analises, mi_lrlslim,
3312  5      mi_zeigs, mi_doas, mi_estype, mi_parp
3313  unlock (unit=index_unit)
3314  if ( status .eq. FOR$IOS_SPERECLC ) then
3315  type *, 'Record #', mi_pkey, ' locked. Waiting 1 sec.', uparrow
3316  call lib$wait( 1.0 )
3317  type *, spaces, uparrow
3318  go to 60410
3319  else if ( ( status .ne. 0 )
3320  1      .or. ( index( 'ER', mi_stat ) .le. 0 ) ) then
3321  go to 60000

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3322         end if
3323     c
3324     c         Experiment for which to plot the DOA spectrum has primary key
3325     c value mi_pkey, and it is a valid DOA spectrum candidate. Determine the
3326     c location of the antenna array description, according to whether it is
3327     c a "descendant" or an "ancestor" experiment.
3328     c
3329     c         if ( mi_park .eq. -1 ) then                ! "Ancestor".
3330     c             selexp = mi_pkey
3331     c         else                ! "Descendant".
3332     c             selexp = mi_park
3333     c         continue
3334     c         read (unit=index_unit,keyeq=selexp,keyid=0,iostat=status)
3335     c             selexp, tch, selpar
3336     c         1
3337     c         unlock (unit=index_unit)
3338     c         if ( status .eq. FOR$IOS_SPERECLOC ) then
3339     c             type *, 'Record #', selexp, ' locked. Waiting 1 sec.', uparrow
3340     c             call lib$wait( 1.0 )                ! Wait 1 second.
3341     c             type *, spaces, uparrow
3342     c             go to 60440
3343     c         else if ( status .ne. 0 ) then                ! Error reading index.
3344     c             mi_pkey = selexp
3345     c             go to 60700
3346     c         end if
3347     c         if ( selpar .ne. -1 ) then
3348     c             selexp = selpar
3349     c             go to 60440
3350     c         end if
3351     c
3352     c         Now create the required Virtual Arrays.
3353     c
3354     c         call init_va
3355     c         ants_by_2(1) = mi_ants
3356     c         ants_by_2(2) = 2
3357     c         ants_lin(1) = mi_ants
3358     c         ants_sqr(1) = mi_ants
3359     c         ants_sqr(2) = mi_ants
3360     c
3361     c         call open_va( ant_add, ants_by_2, 2, dr_type, vstatus, msg )
3362     c         call exit_va( vstatus, msg )
3363     c         call open_va( eig_add, ants_lin, 1, dr_type, vstatus, msg )
3364     c         call exit_va( vstatus, msg )
3365     c         call open_va( eigv_add, ants_sqr, 2, dc_type, vstatus, msg )
3366     c         call exit_va( vstatus, msg )
3367     c
3368     c         Retrieve the antenna array description matrix and the
3369     c         eigenvalue problem solutions from the appropriate intermediate files.
3370     c
3371     c         assign 20000 to err_return
3372     c
3373     c         work_file = 'MUSIC_INPUT_____DAT'
3374     c         write (unit=work_file( 12 : 21 ),fmt=90020,iostat=status,err=60510)
3375     c             selexp
3376     c
3377     c         open (unit=data_unit,file='6000$RT:LCHUCK.PARAMS1'//work_file,
3378     c             status='OLD',organization='SEQUENTIAL',access='SEQUENTIAL',

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3379      2      form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
3380      3      READONLY,iostat=status,err=60560)
3381      read (unit=data_unit,iostat=status,err=60500)
3382      if ( .not. (
3383      1      readarr( %val( ant_add ), dr_type, mi_ants, 2, data_unit, 32 )
3384      2      ) ) go to 60500
3385      close (unit=data_unit)
3386
3387      c
3388      work_file( 7 : 11 ) = 'EIGVS'
3389      write (unit=work_file( 12 : 21 ),fmt=90020,iostat=status,err=60510)
3390      1      mi_pkey
3391      open (unit=data_unit,file='6000%RT:[CHUCK.PARAMS]//work_file,
3392      1      status='OLD',organization='SEQUENTIAL',access='SEQUENTIAL',
3393      2      form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
3394      3      READONLY,iostat=status,err=60560)
3395      do iemp = 1, 5
3396      1      if ( .not. ( skiparr( mi_ants, mi_ants, data_unit, 16 ) ) )
3397      2      go to 60500
3398      end do
3399      if ( ( mi_park .eq. -1 ) .and. ( mi_which .gt. 0 ) ) then
3400      1      if ( .not. ( skiparr( mi_ants, mi_ants, data_unit, 16 ) ) )
3401      2      go to 60500
3402      end if
3403      if ( .not. (
3404      1      readarr( %val( eig_add ), dr_type, mi_ants, 1, data_unit, 32 )
3405      2      ) ) go to 60500
3406      if ( .not. (
3407      1      readarr( %val( eigv_add ), dc_type, mi_ants, mi_ants,
3408      2      data_unit, 16 ) ) ) go to 60500
3409      close (unit=data_unit)
3410
3411      c
3412      Now open scratch file for DDA storage, plot DDA spectra and
3413      allow peak selection.
3414      c
3415      open (unit=data_unit,file='6000%RT:[CHUCK.PARAMS]SCRATCH.DAT',
3416      1      form='UNFORMATTED',recordtype='FIXED',recl=2,
3417      2      organization='RELATIVE',access='DIRECT',
3418      3      status='NEW',dispose='DELETE',iostat=status,err=60830)
3419
3420      c
3421      psel = 'C'
3422      do while ( psel .eq. 'C' ) ! Until operator tires of DDA finding.
3423      c
3424      Accept number of "noise" eigenvectors to use.
3425      c
3426      num_zeigs = type_eigs( %val( eig_add ), mi_ants, mi_to! )
3427      if ( num_zeigs .lt. 0 ) then ! For 0 eigenvalue.
3428      close (unit=data_unit)
3429      call dump_va( vstatus, msg )
3430      call exit_va( vstatus, msg )
3431      mi_stat = 'X'
3432      go to 60600
3433      end if
3434
3435      c
3436      Get range of p values to cover.
3437      c
3438      call prange( mi_begp, mi_endp, mi_delp, fskip, count, iskip )
3439      num_doas = 0

```

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3436 write (unit=data_unit,rec=1,iostat=status,err=60820)
3437 1 num_doas ! Initialize # DOAs found.
3438 c
3439 60450 continue ! Branch for replot selected.
3440 c
3441 c Plot spectra and pick some peaks.
3442 c
3443 if ( music_plot_spect( %val( ant_add ), mi_ants, mi_wlen,
3444 1 %val( eigv_add ), num_zeigs,
3445 2 mi_begp, mi_delp, fskip, count, lskip )
3446 3 ) go to 60800
3447 psel = pick_music_peak( %val( ant_add ), mi_ants, mi_wlen,
3448 1 %val( eigv_add ), num_zeigs,
3449 2 data_unit, mi_begp, mi_delp,
3450 3 fskip, count, lskip )
3451 if ( psel .eq. 'X' ) go to 60800
3452 if ( psel .eq. 'R' ) go to 60450 ! Replot selected.
3453 c
3454 c Peaks selected, initiate new experiments using peaks as DOAs if
3455 c that is the operator's whim.
3456 c
3457 60480 continue
3458 read (unit=data_unit,rec=1,iostat=status,err=60820) num_doas
3459 if ( num_doas .gt. 0 ) then
3460 type 90050, 'Do you wish to start', count,
3461 1 ' new experiments with', num_doas,
3462 2 ' DOAs each ([y]/n)? '
3463 accept 90010, tch
3464 if ( cap( tch ) .ne. 'N' ) then
3465 c
3466 c Select type of estimation for the new experiments.
3467 c
3468 60460 continue
3469 type *, 'Use Enumeration, Sum-of-Powers, Compensated ',
3470 1 'Enumeration,'
3471 type *, 'Sum-of-Powers Loss, Compensated Sum-of-Powers,'
3472 type 90000, ' or Compensated Sum-of-Powers Loss estimation ('
3473 1 // p_etype(estype) // ')? '
3474 read (unit=*,fmt=90010,err=60470) tch
3475 ltemp = index( 'EPNLSC', cap( tch ) )
3476 if ( ltemp .gt. 0 ) estype = ltemp
3477 60470 continue
3478 if ( estype .le. 0 ) go to 60460
3479 c
3480 c Initiated the selected new experiments.
3481 c
3482 doas_lln(1) = num_doas
3483 call open_va( doa_add, doas_lln, 1, dr_type, vstatus, msg )
3484 call exit_va( vstatus, msg )
3485 c
3486 tch = 'D' ! Indicate estimates needed.
3487 tempi = 0 ! No file errors yet.
3488 1 else = 2 ! Start with 1st DOA.
3489 ti_ofile = '6000*RT:[CHUCK.PARAMS]MUSIC_OUTPUT_____DAT'
3490 work_file = 'MUSIC_SAMPL_____DAT'
3491 do i = 1, count
3492 selexp = avibl_prikey( index_unit, 0 )

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3493         if ( selexp .lt. 0 ) then
3494             close (unit=data_unit)
3495             go to 50000
3496         end if
3497         write (unit=ti_ofile( 35 : 44 ),fmt=90020,iostat=status,
3498             1                               err=60820) selexp
3499         write (unit=work_file( 12 : 21 ),fmt=90020,iostat=status,
3500             1                               err=60820) selexp
3501     c
3502     c         Transfer the pertinent DOAs to the appropriate samples file.
3503     c
3504         open (unit=data_unit2,
3505             1         file='6000$RT:[CHUCK.PARAMS]//work_file,status='NEW',
3506             2         organization='SEQUENTIAL',access='SEQUENTIAL',
3507             3         form='UNFORMATTED',recordtype='FIXED',recl=64,
3508             4         dispose='KEEP',iostat=status,err=60820)
3509         call mreaddoas( data_unit, else1, %val( doa_add ), num_doas )
3510         if (.not. (
3511             1         writearr( %val( doa_add ), dr_type, num_doas, 1,
3512             2         data_unit2, 32 ) ) ) go to 60810
3513         close (unit=data_unit2)
3514     c
3515     c         Record the existence of the new experiment.
3516     c
3517         write (unit=index_unit,err=60820)
3518             1         selexp, tch, mi_pkey, tempi, tempi, tempi, mi_dtim,
3519             2         mi_ants, mi_waves, mi_noises, mi_smp, mi_sint, mi_begp,
3520             3         mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen, mi_tol,
3521             4         mi_which, ti_ofile, mi_noises, mi_irlslim, num_zeigs,
3522             5         num_doas, estype, mi_begp +
3523             6         ( fskip + ( i - 1 ) * ( iskip + 1 ) ) * mi_delp
3524         unlock (unit=index_unit)
3525     end do
3526     call free_va( doa_add, vstatus, msg )
3527     call exit_va( vstatus, msg )
3528     end if
3529     c
3530     c         All selected new experiments started. Ask for further
3531     c         instructions.
3532     c
3533         type *, 'Do you wish to start additional new experiments ',
3534             1         'with current DOAs,'
3535         type 90000, ' determine more DOAs, or stop DOA processing '
3536             1         // 'for this experiment (N/D/[X])? '
3537         read (unit=*,fmt=90010,err=60490) tch
3538         tch = cap( tch )
3539         if ( tch .eq. 'D' ) then
3540             psel = 'C'
3541         else if ( tch .eq. 'N' ) then
3542             go to 60480
3543         else
3544             psel = 'X'
3545         end if
3546     end if
3547     end do
3548     60490 continue ! Finished DOA processing for this experiment.
3549     c

```


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3607          accept 90010, answer
3608          read (unit=answer,fmt=*,err=62430) mi_pkey
3609          if ( mi_pkey .lt. 0 ) then
3610      62430          continue                ! Error in accepted #.
3611          mi_pkey
3612          1      = p_index( index_unit, read_unit1, read_unit2, 'ER', .true. )
3613          if ( mi_pkey .eq. -1 ) then        ! No printable experiments.
3614      62000          continue
3615          type *
3616          type *, char( 7 ), 'Can''t print.'
3617          type *
3618          go to 20000
3619          end if
3620          if ( mi_pkey .lt. 0 ) go to 20000  ! Operator cancelled.
3621          end if
3622      c
3623          call print_expm( mi_pkey )
3624          go to 20000
3625      c
3626      c!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!
3627      c
3628      c          Compute performance measures selected.
3629      c
3630          else if ( psel .eq. 'C' ) then
3631          assign 20000 to err_return
3632      c
3633          type *, 'Enter number of the experiment for which to evaluate ',
3634          1      'performance,'
3635          type 90000, '          <CR> to see a list of applicable '
3636          1      // 'experiments: '
3637          accept 90010, answer
3638          read (unit=answer,fmt=*,err=63430) mi_pkey
3639          if ( mi_pkey .lt. 0 ) then
3640      63430          continue                ! Error in accepted #.
3641          mi_pkey
3642          1      = p_index( index_unit, read_unit1, read_unit2, 'IEDR', .true. )
3643          if ( mi_pkey .eq. -1 ) then        ! No valid experiments.
3644      63000          continue
3645          type *
3646          type *, char( 7 ), 'Can''t compute performance measures.'
3647          type *
3648          go to 20000
3649          end if
3650          if ( mi_pkey .lt. 0 ) go to 20000  ! Operator cancelled.
3651          end if
3652      c
3653      63410          continue
3654          read (unit=index_unit,keyeq=mi_pkey,keyid=0,iostat=status)
3655          1      mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
3656          2      mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
3657          3      mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
3658          4      mi_tol, mi_which, mi_ofile, mi_anois, mi_trislim,
3659          5      mi_zeigs, mi_doas, mi_estype, mi_parp
3660          unlock (unit=index_unit)
3661          if ( status .eq. FOR$IOS_SPERECLOC ) then
3662          type *, 'Record #', mi_pkey, ' locked. Waiting 1 sec.', uparrow
3663          call lib$wait( 1.0 )                ! Wait 1 second.

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3664         type *, spaces, uparrow
3665         go to 63410
3666     else if ( ( status .ne. 0 )
3667 1         .or. ( mi_park .eq. -1 )
3668 2         .or. ( index( 'IEDR', mi_stat ) .le. 0 ) ) then
3669         go to 63000
3670     end if
3671 c
3672 c         Experiment for which to evaluation performance has primary key
3673 c value mi_pkey, and it is a valid candidate. Create required virtual
3674 c arrays.
3675 c
3676 c         call init_va
3677 c waves_lin(1) = mi_waves
3678 c
3679 c         call open_va( wave_add, waves_lin, 1, dr_type, vstatus, msg )
3680 c         call exit_va( vstatus, msg )
3681 c         call open_va( exnm_add, waves_lin, 1, di_type, vstatus, msg )
3682 c         call exit_va( vstatus, msg )
3683 c
3684 c         Initialize counters, open work file, and collect DOA estimates.
3685 c
3686 c         p           = mi_parp           ! Value for P matrix estimate.
3687 c         num_acc     = 0                 ! No collected estimates.
3688 c         num_exps    = 0                 ! No experiments used yet.
3689 c         selexp     = mi_pkey           ! For first sample file sel.
3690 c         itemp       = 0                 ! For find file call.
3691 c         work_file   = 'MUSIC_STDOA_____DAT'
3692 c         write (unit=work_file( 12 : 21 ),fmt=90020, iostat=status,err=63830)
3693 c         selexp
3694 c         if ( lib$find_file( '6000*RT:[CHUCK.PARAMS]//work_file,
3695 1         msg, itemp ) ) then           ! File exists.
3696 c         if ( .not. lib$find_file_end( itemp ) ) then ! Error.
3697 63440 c         continue
3698 c         call dump_va( vstatus, msg )
3699 c         call exit_va( vstatus, msg )
3700 c         go to 63000
3701 c         end if
3702 c         type *, char( 7 )
3703 c         type *, 'Performance evaluation control file exists for this ',
3704 1         'experiment.'
3705 c         type *, 'This may indicate that performance evaluation is in ',
3706 1         'progress.'
3707 c         type 90000, ' Do you wish to delete the file (Y/[N])? '
3708 c         accept 90010, tch
3709 c         if ( cap( tch ) .ne. 'Y' ) go to 63440
3710 c         type *, char( 7 )
3711 c         type 90000, ' Are you sure (Y/[N])? '
3712 c         if ( cap( tch ) .ne. 'Y' ) go to 63440
3713 c         call lib$delete_file( '6000*RT:[CHUCK.PARAMS] // work_file
3714 1         // ',*')
3715 c         else
3716 c         if ( .not. lib$find_file_end( itemp ) ) go to 63440 ! Error.
3717 c         end if
3718 c         open (unit=data_unit,file='6000*RT:[CHUCK.PARAMS]//work_file,
3719 1         status='NEW',organization='SEQUENTIAL',access='SEQUENTIAL',
3720 2         form='UNFORMATTED',recordtype='FIXED',recl=8,dispose='KEEP',

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3721      3      iostat=status,err=63830)
3722      work_file( 7 : 11 ) = 'SAMPL'
3723      do while ( mi_park .ge. 0 )      ! Collect DOA estimates.
3724      doas_lin(1) = mi_doas
3725      call open_va( doa_add, doas_lin, 1, dr_type, vstatus, msg )
3726      call exit_va( vstatus, msg )
3727      write (unit=work_file( 12 : 21 ),fmt=90020,iostat=status,err=63820)
3728      1      selexp
3729      c
3730      open (unit=data_unit2,file='6000$RT:[CHUCK.PARAMS]//work_file,
3731      status='OLD',organization='SEQUENTIAL',access='SEQUENTIAL',
3732      form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
3733      2      READONLY,iostat=status,err=60550)
3734      3      if ( .not. ( readarr( %val( doa_add ), dr_type, mi_doas, 1,
3735      data_unit2, 92 ) ) ) go to 63810
3736      1      close (unit=data_unit2)
3737      c
3738      if ( .not. collect_doas( mi_waves, num_acc, num_exps,
3739      1      %val( wave_add ), %val( exnm_add ),
3740      2      data_unit, mi_doas, %val( doa_add ),
3741      3      selexp, mi_zeigs, mi_estype, mi_parp )
3742      4      ) go to 63820
3743      call free_va( doa_add, vstatus, msg )
3744      call exit_va( vstatus, msg )
3745      c      Go up one in ancestor chain.
3746      63420      continue
3747      read (unit=index_unit,keyeq=mi_park,keyld=0,iostat=status)
3748      1      selexp, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
3749      2      mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
3750      3      mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
3751      4      mi_tol, mi_which, mi_ofile, mi_anoises, mi_lrislim,
3752      5      mi_zeigs, mi_doas, mi_estype, mi_parp
3753      unlock (unit=index_unit)
3754      if ( status .eq. FOR$IOS_SPERECL0C ) then
3755      type *, 'Record #', mi_park, ' locked. Waiting 1 sec.', uparrow
3756      call lib$wait( 1.0 )      ! Wait 1 second.
3757      type *, spaces, uparrow
3758      go to 63420
3759      else if ( status .ne. 0 ) then
3760      go to 63820
3761      end if
3762      end do
3763      c
3764      c      DOA estimates for evaluation are all collected. Identify
3765      c      observation and noise sample matrices, and spawn performance measure
3766      c      computation job.
3767      c
3768      if ( mi_waves .gt. num_acc ) then      ! Not enough estimates.
3769      close (unit=data_unit)
3770      call dump_va( vstatus, msg )
3771      call exit_va( vstatus, msg )
3772      type *
3773      type *, 'Not all arriving waves have been identified.'
3774      go to 63000
3775      end if
3776      c
3777      63450      continue

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3778 type *, 'Use Enumeration, Sum-of-Powers, Compensated Enumeration,'
3779 type *, 'Sum-of-Powers Loss, Compensated Sum-of-Powers,'
3780 type 90000, ' or Compensated Sum-of-Powers Loss estimation ('
3781 1 // p_estype(estype) // '?'? '
3782 read (unit=*,fmt=90010,err=63460) tch
3783 itemp = index( 'EPNLSC', cap( tch ) )
3784 if ( itemp .gt. 0 ) estype = itemp
3785 63460 continue
3786 if ( estype .le. 0 ) go to 63450
3787 c
3788 write (unit=data_unit,err=63820)
3789 1 mi_park, mi_pkey, selexp, num_exps, estype, p
3790 if ( .not. ( writearr( %val( wave_add ), dr_type,
3791 1 mi_waves, 1, data_unit, 4 ) ) ) go to 63820
3792 if ( .not. ( writearr( %val( exnm_add ), di_type,
3793 1 mi_waves, 1, data_unit, 8 ) ) ) go to 63820
3794 close (unit=data_unit)
3795 c
3796 call submit_job( spawn_unit, 4 )
3797 call dump_va( vstatus, msg )
3798 call exit_va( vstatus, msg )
3799 go to 20000
3800 c
3801 c File I/O error - performance evaluation.
3802 c
3803 63810 continue
3804 close (unit=data_unit2)
3805 c
3806 63820 continue
3807 close (unit=data_unit)
3808 c
3809 63830 continue
3810 call dump_va( vstatus, msg )
3811 call exit_va( vstatus, msg )
3812 c
3813 63840 continue
3814 type *, char( 7 )
3815 type *, 'Performance evaluation: Error accessing data files.'
3816 type *, 'Experiment #', mi_pkey, ' Performance evaluation ',
3817 1 'abandoned.'
3818 type *, char( 7 )
3819 go to 20000
3820 c
3821 c!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!
3822 c
3823 c Mark active experiment for deletion selected.
3824 c
3825 else if ( psel .eq. 'M' ) then
3826 assign 20000 to err_return
3827 c
3828 type *, 'Enter number of the experiment to mark for deletion,'
3829 type 90000, ' (CR) to see a list of deletable experiments: '
3830 accept 90010, answer
3831 read (unit=answer,fmt=*,err=81430) mi_pkey
3832 if ( mi_pkey .lt. 0 ) then
3833 81430 continue ! Error in accepted #.
3834 mi_pkey

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3835      1      = p_index( index_unit, read_unit1, read_unit2, '-ER', .true. )
3836      if ( mi_pkey .eq. -1 ) then          ! No deletable experiments.
3837      81000      continue
3838      type *
3839      type *, char( 7 ), 'Can't mark for deletion.'
3840      type *
3841      go to 20000
3842      end if
3843      if ( mi_pkey .lt. 0 ) go to 20000      ! Operator cancelled.
3844      end if
3845      c
3846      81410      continue
3847      read (unit=index_unit,keyeq=mi_pkey,keyid=0,iostat=status)
3848      1      mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
3849      2      mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
3850      3      mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
3851      4      mi_tol, mi_which, mi_ofile, mi_anoises, mi_irlslim,
3852      5      mi_zeigs, mi_doas, mi_estype, mi_parp
3853      unlock (unit=index_unit)
3854      if ( status .eq. FOR$IOS_SPERECLOC ) then
3855      type *, 'Record #', mi_pkey, ' locked. Waiting 1 sec.', uparrow
3856      call lib$wait( 1.0 )                  ! Wait 1 second.
3857      type *, spaces, uparrow
3858      go to 81410
3859      else if ( ( status .ne. 0 )
3860      1      .or. ( index( '-ER', mi_stat ) .le. 0 ) ) then
3861      go to 81000
3862      end if
3863      c
3864      c      Experiment to be marked for deletion has primary key value
3865      c      mi_pkey, and it is a valid candidate.
3866      c
3867      type *
3868      type 90030, char( 7 ) // 'Mark experiment #', mi_pkey,
3869      1      ' for DELETION?'
3870      type 90000, ' (Y/[N])? '
3871      accept 90010, answer
3872      psel = cap( answer( 1 : 1 ) )
3873      c
3874      if ( psel .eq. 'Y' ) then
3875      type 90000, ' ' // char( 7 ) // 'Are you sure (Y/[N])? '
3876      accept 90010, answer
3877      psel = cap( answer( 1 : 1 ) )
3878      if ( psel .ne. 'Y' ) go to 81520
3879      c
3880      c      Operator has given assurance that he wants the selected
3881      c      experiment MARKED FOR DELETION.
3882      c
3883      mi_stat = 'X'
3884      type *
3885      type 90030, char( 7 ) // 'Marking experiment #', mi_pkey,
3886      1      ' for DELETION!'
3887      type *
3888      go to 60600
3889      else
3890      81520      continue
3891      type *

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3892         type 90030, 'Experiment #', mi_pkey, ' not altered.'
3893     type *
3894     end if
3895     go to 20000
3896 c
3897 c!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!
3898 c
3899 c             Delete completed or failed experiment selected.
3900 c
3901     else if ( psel .eq. 'K' ) then
3902         assign 20000 to err_return
3903 c
3904     type *, 'Enter number of the experiment for which to DELETE files,'
3905     type 90000, '                               <CR> to see a list of deletable '
3906     1                                           // 'experiments: '
3907     accept 90010, answer
3908     read (unit=answer,fmt=*,err=80430) mi_pkey
3909     if ( mi_pkey .lt. 0 ) then
3910 80430         continue                               ! Error in accepted #.
3911         mi_pkey
3912     1         = p_index( index_unit, read_unit1, read_unit2, '-RX', .true. )
3913         if ( mi_pkey .eq. -1 ) then               ! No deletable experiments.
3914 80000         continue
3915             type *
3916             type *, char( 7 ), 'Can''t delete.'
3917             type *
3918             go to 20000
3919         end if
3920         if ( mi_pkey .lt. 0 ) go to 20000         ! Operator cancelled.
3921     end if
3922 c
3923 80410     continue
3924         read (unit=index_unit,keyeq=mi_pkey,keyid=0,lostat=status)
3925     1         mi_pkey, mi_stat, mi_park, mi_lerr, mi_eerr, mi_serr,
3926     2         mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
3927     3         mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
3928     4         mi_tol, mi_which, mi_ofile, mi_anaises, mi_irislim,
3929     5         mi_zeigs, mi_doas, mi_estype, mi_parp
3930     unlock (unit=index_unit)
3931     if ( status .eq. FOR$IOS_SPERECL0C ) then
3932         type *, 'Record #', mi_pkey, ' locked. Waiting 1 sec.', uparrow
3933         call lib$wait( 1.0 )                       ! Wait 1 second.
3934         type *, spaces, uparrow
3935         go to 80410
3936     else if ( ( status .ne. 0 )
3937     1         .or. ( index( '-RX', mi_stat ) .le. 0 ) ) then
3938         go to 80000
3939     end if
3940 c
3941 c             Experiment to delete has primary key value mi_pkey, and it is a
3942 c             valid deletion candidate.
3943 c
3944     type *
3945     type *, char( 7 ), 'Delete ALL intermediate files ',
3946     1                                           'AND output file AND index entry ',
3947     2                                           'for experiment'
3948     type 90030, '#', mi_pkey, '?'

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3949      type 90000, ' (Y/[NJ]? '
3950      accept 90010, answer
3951      psel = cap( answer( 1 : 1 ) )
3952      c
3953      if ( psel .eq. 'Y' ) then
3954          type 90000, ' ' // char( 7 ) // 'Are you sure (Y/[NJ]? '
3955          accept 90010, answer
3956          psel = cap( answer( 1 : 1 ) )
3957          if ( psel .ne. 'Y' ) go to 80520
3958      c
3959      80530      continue
3960      read (unit=index_unit,key=mi_pkey,keyid=0,lostat=status)
3961      if      ( status .eq. FOR$IOS_SPERECLC ) then
3962          unlock (unit=index_unit)
3963          type *, 'Index record locked by another program ...'
3964          call lib$wait( 10.0 )           ! Wait 10 seconds.
3965          type *, 'Trying again.'
3966          go to 80530
3967      else if ( status .ne. 0 ) then
3968          unlock (unit=index_unit)
3969          go to 60700
3970      end if
3971      c
3972      c           Operator has given assurance that he wants all files associated
3973      c           with the selected experiment DELETED.
3974      c
3975      work_file = 'MUSIC_*_____DAT'
3976      write (unit=work_file( 8 : 17 ),fmt=90020,err=80540) mi_pkey
3977      call lib$delete_file( '6000*RT:[CHUCK.PARAMS]' // work_file
3978      1 // ',*' )
3979      delete (unit=index_unit,lostat=status,err=60700)
3980      unlock (unit=index_unit)
3981      c
3982      type *
3983      type 90030, char( 7 ) // 'ALL files for experiment #', mi_pkey,
3984      1 // ' DELETED!'
3985      type *
3986      else
3987      80520      continue
3988      type *
3989      type 90030, 'Experiment #', mi_pkey, ' not altered.'
3990      type *
3991      end if
3992      go to 20000
3993      c
3994      80540      continue
3995      type *
3996      type 90030, char( 7 ) // 'Unable to form data file name for '
3997      1 // 'experiment #', mi_pkey, '...'
3998      go to 80520
3999      c
4000      c!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!
4001      c
4002      c           Results display for active experiment selected. Plots
4003      c           available are:
4004      c           1)      Picture of radar array with arriving waves.
4005      c           2)      Elements of the signal in space, observation, noise, !

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4006      c                                Lp signal in space estimate, and residual      !
4007      c                                vectors.                                     !
4008      c                                3)   MUSIC direction-of-arrival (DOA) spectra and  !
4009      c                                sharpness measures.                         !
4010      c                                !                                         !
4011      c                                else if ( psel .eq. 'L' ) then                !
4012      c                                assign 61500 to err_return                    !
4013      c                                !                                         !
4014      c                                vsel = 1                                    !
4015      c                                do while ( vsel .ne. 9 )                    !
4016      c                                type *                                       !
4017      c                                if ( oinst_shown ) then                       !
4018      c                                if ( p_index( index_unit, read_unit1, read_unit2,  !
4019      c                                'ER', .false. ) .ge. 0 ) then                 !
4020      c                                type *, 'c = P, S, C, O, N, D, I, A, X, or ?.'    !
4021      c                                else if ( p_index( index_unit, read_unit1, read_unit2,  !
4022      c                                'ID', .false. ) .ge. 0 ) then                 !
4023      c                                type *, 'c = P, S, C, O, N, I, X, or ?.'    !
4024      c                                else                                           !
4025      c                                type *, 'P, X, or ?.'                          !
4026      c                                end if                                         !
4027      c                                else                                           !
4028      c                                10420 continue                               !
4029      c                                oinst_shown = .true.                          !
4030      c                                if ( p_index( index_unit, read_unit1, read_unit2,  !
4031      c                                'IEDR', .false. ) .ge. 0 ) then                 !
4032      c                                type *, 'Selections are of the form c nnn, where c is:'  !
4033      c                                type *, '      P(p): array picture;'          !
4034      c                                else                                           !
4035      c                                type *, 'Enter P or p to plot array picture, '    !
4036      c                                'X or x to stop plotting.'                    !
4037      c                                go to 61440                                    !
4038      c                                end if                                         !
4039      c                                if ( p_index( index_unit, read_unit1, read_unit2,  !
4040      c                                'IEDR', .false. ) .ge. 0 ) then                 !
4041      c                                type *, '      S(s): signal-in-space (or estimate) entry;'  !
4042      c                                type *, '      C(c): noise-free observation vector entry ',  !
4043      c                                '      ("ancestors" only);'                    !
4044      c                                type *, '      O(o): noisy observation (or residual) entry;'  !
4045      c                                type *, '      N(n): noise vector entry ("ancestors" only);'  !
4046      c                                type *, '      I(i): all "input" vector entries (S, C, O, N);'  !
4047      c                                end if                                         !
4048      c                                if ( p_index( index_unit, read_unit1, read_unit2,  !
4049      c                                'ER', .false. ) .ge. 0 ) then                 !
4050      c                                type *, '      D(d): direction of arrival spectra;'    !
4051      c                                type *, '      A(a): all trial vector entries (O, N, D);'  !
4052      c                                end if                                         !
4053      c                                if ( p_index( index_unit, read_unit1, read_unit2,  !
4054      c                                'IEDR', .false. ) .ge. 0 ) then                 !
4055      c                                type *, '      X(x): end pictures;'                !
4056      c                                type *, '      and nnn is an integer (<= 3 digits) choosing ',  !
4057      c                                '      the vector entry # to plot'            !
4058      c                                '      (applies only to selections S, C, O, and N).. '  !
4059      c                                else                                           !
4060      c                                type *, '      X(x): end pictures.'                !
4061      c                                end if                                         !
4062      c                                61440 continue                               !

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4063          type *
4064          type *, 'Note: Entering ? repeats this explanation.'
4065          end if
4066      c
4067          type *
4068          type 90000, ' Please select the activity to perform (cnnn): '
4069          accept 90010, answer
4070          chsel = answer( 2 : 4 )
4071          vsel = index( 'PSCONIDAX', cap( answer( 1 : 1 ) ) )
4072          if ( vsel .le. 0 ) go to 10420
4073          if ( vsel .eq. 9 ) go to 61500
4074          type *, ' For what experiment # shall '
4075          1          // ptypes(vsel) // ' be computed?'
4076          type 90000, '          <<CR> for a list of applicable '
4077          1          // 'experiments): '
4078      c
4079          accept 90010, answer
4080          read (unit=answer,fmt=*,err=61430) mi_pkey
4081          if ( mi_pkey .lt. 0 ) then
4082      61430          continue          ! Error in accepted #.
4083          mi_pkey = p_index( index_unit, read_unit1, read_unit2,
4084          1          ptstat(vsel), .true. )
4085          if ( mi_pkey .eq. -1 ) then          ! No outputable exp's.
4086      61010          continue
4087          type *
4088          type *, char( 7 ), 'Can't plot.'
4089          type *
4090          go to 61500
4091          end if
4092          if ( mi_pkey .lt. 0 ) go to 61500          ! Operator cancelled.
4093          end if
4094      c
4095      61410          continue
4096          read (unit=index_unit, keyeq=mi_pkey, keyid=0, lostat=status)
4097          1          mi_pkey, mi_stat, mi_park, mi_lerr, mi_eerr, mi_serr,
4098          2          mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
4099          3          mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
4100          4          mi_tol, mi_which, mi_ofile, mi_anoises, mi_lrlslim,
4101          5          mi_zelgs, mi_doas, mi_estype, mi_parp
4102          unlock (unit=index_unit)
4103          if ( status .eq. FOR#IOS_SPERECLC ) then
4104          type *, 'Record #', mi_pkey, ' locked. Waiting 1 sec.', uparrow
4105          call lib$wait( 1.0 )          ! Wait 1 second.
4106          type *, spaces, uparrow
4107          go to 61410
4108          else if ( ( status .ne. 0 )
4109          1          .or. ( index( ptstat(vsel), mi_stat ) .le. 0 ) ) then
4110          go to 61010
4111          end if
4112      c
4113      c          Experiment for which to display results has primary key value
4114      c          mi_pkey, and it is a valid results display candidate.
4115      c
4116          if ( .not. pinfo_shown ) then
4117      61090          continue
4118          pinfo_shown = .true.
4119          type *

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4120         type *, '                                PICTURES'
4121         type *
4122         type *, 'During plotting, the computer types an appropriate ',
4123         1         "ready to plot" message,'
4124         type *, 'then waits for a <CR> before proceeding with the ',
4125         1         'plot. When a plot is'
4126         type *, 'finished, the computer waits for another <CR> before ',
4127         1         'continuing with the next'
4128         type *, 'program step.'
4129         type *
4130     end if
4131     c
4132     c         Now plot the selected results.
4133     c
4134     c         .....Operator selected picture of antenna array.....
4135     c
4136     c         if      ( vsel .eq. 1 ) then                ! Picture chosen.
4137     c
4138     c         Find antenna array description, which may be part of the
4139     c         selected experiment's "ancestor" experiment.
4140     c
4141     c         if ( mi_park .eq. -1 ) then                  ! "Ancestor".
4142     c             selexp = mi_pkey
4143     c         else                                        ! "Descendant".
4144     c             selexp = mi_park
4145     c             continue
4146     c             read (unit=index_unit,keyeq=selexp,keyld=0,lostat=status)
4147     c             1         selexp, tch, selpar
4148     c             unlock (unit=index_unit)
4149     c             if      ( status .eq. FOR$IOS_SPERECLOC ) then
4150     c                 type *, 'Record #', selexp, ' locked. Waiting 1 sec.',
4151     c                 1         uparrow
4152     c                 call lib$wait( 1.0 )                ! Wait 1 second.
4153     c                 type *, spaces, uparrow
4154     c                 go to 61470
4155     c             else if ( status .ne. 0 ) then            ! Error reading index.
4156     c                 mi_pkey = selexp
4157     c                 go to 60700
4158     c             end if
4159     c             if ( selpar .ne. -1 ) then
4160     c                 selexp = selpar
4161     c                 go to 61470
4162     c             end if
4163     c         end if
4164     c
4165     c         Now create the required Virtual Arrays.
4166     c
4167     c         call init_va
4168     c         ants_by_2(1) = mi_ants
4169     c         ants_by_2(2) = 2
4170     c         waves_by_9(1) = mi_waves
4171     c         waves_by_9(2) = 9
4172     c
4173     c         call open_va( ant_add, ants_by_2, 2, dr_type, vstatus, msg )
4174     c         call exit_va( vstatus, msg )
4175     c         call open_va( wave_add, waves_by_9, 2, dr_type, vstatus, msg )
4176     c         call exit_va( vstatus, msg )

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DO_MUSIC                                18-Apr-1988 14:00:27    VAX FC
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4177    c
4178        work_file = 'MUSIC_INPUT_____DAT'
4179        write (unit=work_file( 12 : 21 ),fmt=90020,iostat=status,
4180            1          err=60510) selexp
4181        open (unit=data_unit,file='6000$RT:[CHUCK.PARAMS]//work_file,
4182            1          status='OLD',organization='SEQUENTIAL',
4183            2          access='SEQUENTIAL',form='UNFORMATTED',
4184            3          recordtype='FIXED',recl=64,dispose='KEEP',
4185            4          READONLY,iostat=status,err=60560)
4186        read (unit=data_unit,err=60500)
4187        if ( .not. (
4188            1          readarr( %val( ant_add ), dr_type, mi_ants, 2,
4189            2          data_unit, 32 ) ) ) go to 60500
4190        if ( .not. (
4191            1          readarr( %val( wave_add ), dr_type, mi_waves, 9,
4192            2          data_unit, 32 ) ) ) go to 60500
4193        close (unit=data_unit)
4194    c
4195        call music_picture( %val( ant_add ), mi_ants,
4196            1          %val( wave_add ), mi_waves )
4197    c
4198        call dump_va( vstatus, msg )
4199        call exit_va( vstatus, msg )
4200    c
4201    c.....Operator selected plot(s) of input vector(s).....
4202    c
4203        else if ( vsel .le. 6 ) then
4204            if ( vsel .ne. 6 ) then
4205                else = -1
4206                read(unit=chsel,fmt=*,err=61200) else1
4207            end if
4208            61200    continue
4209    c
4210        ancestor = ( mi_park .eq. -1 )    ! True for "ancestor" exper.
4211        if ( ancestor ) then            ! "Ancestor".
4212            wavs_by_smp(1) = mi_waves
4213            ants_by_smp(1) = mi_ants
4214            ants_by_smp(2) = mi_smp
4215        else                            ! "Descendant".
4216            if ( ( vsel .eq. 3 ) .or. ( vsel .eq. 5 ) ) go to 61010
4217    c
4218    c        Get number of accumulated DOAs including this experiment.
4219    c
4220        num_doas = mi_doas
4221        selexp = mi_park
4222    62470    continue
4223        read (unit=index_unit,keyeq=selexp,keyid=0,iostat=status)
4224            1          selexp, tch, selpar, mi_ierr, mi_eerr, mi_serr,
4225            2          mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
4226            3          mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
4227            4          mi_tol, mi_which, mi_ofile, mi_anoises, mi_irlslm,
4228            5          mi_zelgs, itemp, mi_estype, mi_parp
4229        unlock (unit=index_unit)
4230        if ( status .eq. FOR$IOS_SPERECLOC ) then
4231            type *, 'Record #', selexp, ' locked. Waiting 1 sec.',
4232            1          uparrow
4233        call lib$wait( 1.0 )            ! Wait 1 second.

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4234         type *, spaces, uparrow
4235         go to 62470
4236         else if ( status .ne. 0 ) then           ! Error reading index.
4237             mi_pkey = selexp
4238             go to 60700
4239         end if
4240         if ( selpar .ne. -1 ) then
4241             num_doas = num_doas + ltemp
4242             selexp = selpar
4243             go to 62470
4244         end if
4245     c
4246         wavs_by_smp(1) = num_doas
4247     end if
4248     wavs_by_smp(2) = mi_smp
4249     smp_by_ants(1) = mi_smp
4250     smp_by_ants(2) = mi_ants
4251     c
4252     call init_va
4253     call open_va( sigvm_add, wavs_by_smp, 2, dc_type, vstatus, msg )
4254     call exit_va( vstatus, msg )
4255     call open_va( obsmat_add, smp_by_ants, 2, dc_type, vstatus, msg )
4256     call exit_va( vstatus, msg )
4257     if ( ancestor ) then           ! "Ancestor".
4258         call open_va( climat_add, ants_by_smp, 2, dc_type, vstatus,
4259             msg )
4260         call exit_va( vstatus, msg )
4261         call open_va( noisemat_add, smp_by_ants, 2, dc_type, vstatus,
4262             msg )
4263         call exit_va( vstatus, msg )
4264     end if
4265     c
4266     work_file = 'MUSIC_SAMPL_____DAT'
4267     write (unit=work_file( 12 : 21 ),fmt=90020,iostat=status,
4268         err=60510) mi_pkey
4269     open (unit=data_unit,file='6000*RT:[CHUCK.PARAMS]//work_file,
4270         status='OLD',organization='SEQUENTIAL',
4271         access='SEQUENTIAL',form='UNFORMATTED',
4272         recordtype='FIXED',recl=64,dispose='KEEP',
4273         READONLY,iostat=status,err=60560)
4274     if ( .not. ancestor ) then           ! "Descendant".
4275         if ( .not. ( skiparr( mi_doas, 1,           ! Skip DOAs.
4276             data_unit, 32 ) ) ) go to 60500
4277     end if
4278     if ( .not. (
4279         readarr( %val( obsmat_add ), dc_type, mi_smp, mi_ants,
4280             data_unit, 16 ) ) ) go to 60500
4281     if ( ancestor ) then           ! "Ancestor".
4282         if ( .not. (
4283             readarr( %val( noisemat_add ), dc_type, mi_smp,
4284                 mi_ants, data_unit, 16 ) ) ) go to 60500
4285     end if
4286     if ( .not. (
4287         readarr( %val( sigvm_add ), dc_type, wavs_by_smp(1),
4288             mi_smp, data_unit, 16 ) ) ) go to 60500
4289     if ( ancestor ) then           ! "Ancestor".
4290         if ( .not. (

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4291      1      readarr( %val( clinmat_add ), dc_type, mi_ants, mi_smp,
4292      2      data_unit, 16 ) ) go to 60500
4293      end if
4294      close (unit=data_unit)
4295      c
4296      call music_plot_inps( vsel, else1,
4297      1      %val( sigvm_add ), %val( clinmat_add ),
4298      2      %val( obsmat_add ), %val( noisemat_add ),
4299      3      mi_which, mi_ants, wavs_by_smp(1), mi_smp,
4300      4      ancestor )
4301      c
4302      call dump_va( vstatus, msg )
4303      call exit_va( vstatus, msg )
4304      c
4305      c.....Operator selected DOA spectrum plot(s).....
4306      c
4307      else if ( vsel .eq. 7 ) then
4308      c
4309      c      Find antenna array description, which may be part of the
4310      c      selected experiment's "ancestor" experiment.
4311      c
4312      if ( mi_park .eq. -1 ) then      ! "Ancestor".
4313      selexp = mi_pkey
4314      else      ! "Descendant".
4315      selexp = mi_park
4316      61450      continue
4317      read (unit=index_unit, keyeq=selexp, keyid=0, iostat=status)
4318      1      selexp, tch, selpar
4319      unlock (unit=index_unit)
4320      if ( status .eq. FOR$IOS_SPERECLC ) then
4321      type *, 'Record #', selexp, ' locked. Waiting 1 sec.',
4322      1      uparrow
4323      call lib$wait( 1.0 )      ! Wait 1 second.
4324      type *, spaces, uparrow
4325      go to 61450
4326      else if ( status .ne. 0 ) then      ! Error reading index.
4327      mi_pkey = selexp
4328      go to 60700
4329      end if
4330      if ( selpar .ne. -1 ) then
4331      selexp = selpar
4332      go to 61450
4333      end if
4334      end if
4335      c
4336      c      Now open required Virtual Arrays.
4337      c
4338      call init_va
4339      ants_by_2(1) = mi_ants
4340      ants_by_2(2) = 2
4341      ants_sqr(1) = mi_ants
4342      ants_sqr(2) = mi_ants
4343      ants_lin(1) = mi_ants
4344      c
4345      call open_va( ant_add, ants_by_2, 2, dr_type, vstatus, msg )
4346      call exit_va( vstatus, msg )
4347      call open_va( eig_add, ants_lin, 1, dr_type, vstatus, msg )

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DO_MUSIC                                     18-Apr-1988 14:00:27   VA
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4343      call exit_va( vstatus, msg )
4349      call open_va( eigv_add, ants_sqr, 2, dc_type, vstatus, msg )
4350      call exit_va( vstatus, msg )
4351      c
4352      work_file = 'MUSIC_INPUT_____DAT'
4353      write (unit=work_file( 12 : 21 ),fmt=90020,iostat=status,
4354             1                               err=60510) selexp
4355      open (unit=data_unit,file='6000*RT:[CHUCK.PARAMS]//work_file,
4356             1                               status='OLD',organization='SEQUENTIAL',
4357             2                               access='SEQUENTIAL',form='UNFORMATTED',
4358             3                               recordtype='FIXED',recl=64,dispose='KEEP',
4359             4                               READONLY,iostat=status,err=60560)
4360      read (unit=data_unit,err=60500)
4361      if ( .not. (
4362             1                               readarr( %val( ant_add ), dr_type, mi_ants, 2,
4363             2                               data_unit, 32 ) ) ) go to 60500
4364      close (unit=data_unit)
4365      work_file( 7 : 11 ) = 'EIGVS'
4366      write (unit=work_file( 12 : 21 ),fmt=90020,iostat=status,
4367             1                               err=60510) mi_pkey
4368      open (unit=data_unit,file='6000*RT:[CHUCK.PARAMS]//work_file,
4369             1                               status='OLD',organization='SEQUENTIAL',
4370             2                               access='SEQUENTIAL',form='UNFORMATTED',
4371             3                               recordtype='FIXED',recl=64,dispose='KEEP',
4372             4                               READONLY,iostat=status,err=60560)
4373      do itemp = 1, 5
4374          if ( .not. ( skiparr( mi_ants, mi_ants, data_unit, 16 ) ) )
4375              1                               go to 60500
4376      end do
4377      if ( ( mi_park .eq. -1 ) .and. ( mi_which .gt. 0 ) ) then
4378          if ( .not. ( skiparr( mi_ants, mi_ants, data_unit, 16 ) ) )
4379              1                               go to 60500
4380      end if
4381      if ( .not. (
4382             1                               readarr( %val( eig_add ), dr_type, mi_ants, 1,
4383             2                               data_unit, 32 ) ) ) go to 60500
4384      if ( .not. (
4385             1                               readarr( %val( eigv_add ), dc_type, mi_ants, mi_ants,
4386             2                               data_unit, 16 ) ) ) go to 60500
4387      close (unit=data_unit)
4388      c
4389      c                               Accept number of "noise" eigenvectors to use.
4390      c
4391      num_zeigs = type_eigs( %val( eig_add ), mi_ants, mi_tol )
4392      if ( num_zeigs .lt. 0 ) then
4393          call dump_va( vstatus, msg )
4394          call exit_va( vstatus, msg )
4395          go to 61500
4396      end if
4397      c
4398      c                               Open scratch file for DOA storage.
4399      c
4400      open (unit=data_unit,file='6000*RT:[CHUCK.PARAMS]SCRATCH.DAT',
4401             1                               form='UNFORMATTED',recordtype='FIXED',recl=2,
4402             2                               organization='RELATIVE',access='DIRECT',
4403             3                               status='NEW',dispose='DELETE',iostat=status,err=60830)
4404      c

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DD_MUSIC

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4405      c          Do plot(s) (finally!).
4406      c
4407      c      psel = 'R'
4408      c      do while ( psel .eq. 'R' )          ! Until operator tires.
4409      c          call prange( mi_begp, mi_endp, mi_delp, fskip, count, iskip )
4410      c          dummy = music_plot_spect( %val( ant_add ), mi_ants, mi_wlen,
4411      c              1          %val( eigv_add ), num_zeigs,
4412      c              2          mi_begp, mi_delp, fskip, count, iskip )
4413      c
4414      c          num_doas = 0
4415      c          write (unit=data_unit,rec=1,iostat=status,err=60820)
4416      c              1          num_doas ! Initialize # DOAs found.
4417      c          psel = pick_music_peak( %val( ant_add ), mi_ants, mi_wlen,
4418      c              1          %val( eigv_add ), num_zeigs,
4419      c              2          data_unit, mi_begp, mi_delp,
4420      c              3          fskip, count, iskip )
4421      c
4422      c      end do
4423      c
4424      c      close (unit=data_unit)
4425      c      call dump_va( vstatus, msg )
4426      c      call exit_va( vstatus, msg )
4427      c.....Operator selected plots of "all" vectors for a trial...
4428      c
4429      c      else
4430      c
4431      c          Find antenna array description, which may be part of the
4432      c      selected experiment's "ancestor" experiment.
4433      c
4434      c          ancestor = ( mi_park .eq. -1 )      ! True for "ancestor" exper.
4435      c          if ( ancestor ) then                ! "Ancestor".
4436      c              selexp = mi_pkey
4437      c          else                                ! "Descendant".
4438      c              selexp = mi_park
4439      c          continue
4440      c      61460      read (unit=index_unit,keyeq=selexp,keyid=0,iostat=status)
4441      c              1          selexp, tch, selpar
4442      c          unlock (unit=index_unit)
4443      c          if ( status .eq. FOR$IOS_SPERECLD ) then
4444      c              type *, 'Record #', selexp, ' locked. Waiting 1 sec.',
4445      c              1          uparrow
4446      c              call lib$wait( 1.0 )          ! Wait 1 second.
4447      c              type *, spaces, uparrow
4448      c              go to 61460
4449      c          else if ( status .ne. 0 ) then      ! Error reading index.
4450      c              mi_pkey = selexp
4451      c              go to 60700
4452      c          end if
4453      c          if ( selpar .ne. -1 ) then
4454      c              selexp = selpar
4455      c              go to 61460
4456      c          end if
4457      c      end if
4458      c
4459      c      Now open required Virtual Arrays.
4460      c
4461      c      call init_va

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4462      ants_lin(1)      = mi_ants
4463      ants_by_2(1)     = mi_ants
4464      ants_by_2(2)     = 2
4465      ants_sqr(1)      = mi_ants
4466      ants_sqr(2)      = mi_ants
4467      if ( ancestor ) then                ! "Ancestor".
4468          ants_by_smp(1) = mi_ants
4469          ants_by_smp(2) = mi_smp
4470      end if
4471      wavs_by_smp(2) = mi_smp
4472      smp_by_ants(1) = mi_smp
4473      smp_by_ants(2) = mi_ants
4474
4475      c
4476      call open_va( ant_add, ants_by_2, 2, dr_type, vstatus, msg )
4477      call exit_va( vstatus, msg )
4478      call open_va( eig_add, ants_lin, 1, dr_type, vstatus, msg )
4479      call exit_va( vstatus, msg )
4480      call open_va( eigv_add, ants_sqr, 2, dc_type, vstatus, msg )
4481      call exit_va( vstatus, msg )
4482      call open_va( obsmat_add, smp_by_ants, 2, dc_type, vstatus, msg )
4483      call exit_va( vstatus, msg )
4484      if ( ancestor ) then                ! "Ancestor".
4485          call open_va( noisemat_add, smp_by_ants, 2, dc_type, vstatus,
4486              msg )
4487          call exit_va( vstatus, msg )
4488          call open_va( climat_add, ants_by_smp, 2, dc_type, vstatus,
4489              msg )
4490          call exit_va( vstatus, msg )
4491      end if
4492
4493      c
4494      work_file = 'MUSIC_INPUT_____DAT'
4495      write (unit=work_file( 12 : 21 ),fmt=90020,iostat=status,
4496          err=60510 ) selexp
4497      open (unit=data_unit,file='6000*RT:[CHUCK.PARAMS]//work_file,
4498          status='OLD',organization='SEQUENTIAL',
4499          access='SEQUENTIAL',form='UNFORMATTED',
4500          recordtype='FIXED',recl=64,dispose='KEEP',
4501          READONLY,iostat=status,err=60560)
4502      read (unit=data_unit,err=60500)
4503      if ( .not. (
4504          readarr( %val( ant_add ), dr_type, mi_ants, 2,
4505              data_unit, 32 ) ) ) go to 60500
4506      close (unit=data_unit)
4507      work_file( 7 : 11 ) = 'EIGVS'
4508      write (unit=work_file( 12 : 21 ),fmt=90020,iostat=status,
4509          err=60510 ) mi_pkey
4510      open (unit=data_unit,file='6000*RT:[CHUCK.PARAMS]//work_file,
4511          status='OLD',organization='SEQUENTIAL',
4512          access='SEQUENTIAL',form='UNFORMATTED',
4513          recordtype='FIXED',recl=64,dispose='KEEP',
4514          READONLY,iostat=status,err=60560)
4515      do ltemp = 1, 5
4516          if ( .not. ( skiparr( mi_ants, mi_ants, data_unit, 16 ) ) )
4517              go to 60500
4518      end do
4519      if ( ( mi_park .eq. -1 ) .and. ( mi_which .gt. 0 ) ) then
4520          if ( .not. ( skiparr( mi_ants, mi_ants, data_unit, 16 ) ) )

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4519          1                                go to 60500
4520      end if
4521      if ( .not. (
4522          1          readarr( %val( eig_add ), dr_type, mi_ants, 1,
4523          2                                data_unit, 32 ) ) ) go to 60500
4524      if ( .not. (
4525          1          readarr( %val( eigv_add ), dc_type, mi_ants, mi_ants,
4526          2                                data_unit, 16 ) ) ) go to 60500
4527      close (unit=data_unit)
4528      work_file( 7 : 11 ) = 'SAMPL'
4529      open (unit=data_unit,file='6000*RT:[CHUCK.PARAMS]//work_file,
4530          1          status='OLD',organization='SEQUENTIAL',
4531          2          access='SEQUENTIAL',form='UNFORMATTED',
4532          3          recordtype='FIXED',recl=64,dispose='KEEP',
4533          4          READONLY,iostat=status,err=60560)
4534      if ( .not. ancestor ) then                                ! "Descendant".
4535          if ( .not. ( skiparr( mi_doas, 1,                                ! Skip DOAs.
4536          1                                data_unit, 32 ) ) ) go to 60500
4537      end if
4538      if ( .not. (
4539          1          readarr( %val( obsmat_add ), dc_type, mi_smp, mi_ants,
4540          2                                data_unit, 16 ) ) ) go to 60500
4541      if ( ancestor ) then                                ! "Ancestor".
4542          if ( .not. (
4543          1          readarr( %val( noisemat_add ), dc_type, mi_smp,
4544          2                                mi_ants, data_unit, 16 ) ) ) go to 60500
4545          if ( .not. ( skiparr( mi_waves, mi_smp, ! Skip sigvm_add.
4546          1                                data_unit, 16 ) ) ) go to 60500
4547          if ( .not. (
4548          1          readarr( %val( clinmat_add ), dc_type, mi_ants, mi_smp,
4549          2                                data_unit, 16 ) ) ) go to 60500
4550      end if
4551      close (unit=data_unit)
4552      c
4553      c          Accept number of "noise" eigenvectors to use.
4554      c
4555      num_zeigs = type_eigs( %val( eig_add ), mi_ants, mi_tol )
4556      if ( num_zeigs .lt. 0 ) then
4557          call dump_va( vstatus, msg )
4558          call exit_va( vstatus, msg )
4559          go to 61500
4560      end if
4561      c
4562      c          At last do plots.
4563      c
4564      call music_plot_trial( %val( obsmat_add ), %val( noisemat_add ),
4565          1          %val( clinmat_add ), mi_ants, mi_smp, mi_begp,
4566          2          mi_endp, mi_delp, %val( ant_add ), mi_wlen,
4567          3          %val( eigv_add ), num_zeigs, mi_which,
4568          4          ancestor )
4569      c
4570      call dump_va( vstatus, msg )
4571      call exit_va( vstatus, msg )
4572      end if
4573      61500      continue
4574      end do
4575      c

```

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4576      c          Operator finished viewing results for the selected experiment
4577      c          return to ask the operator what to do next.
4578      c
4579      c          go to 20000
4580      c
4581      c!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!
4582      c
4583      c          Input file re-creation selected.
4584      c
4585      c          else if ( psel .eq. '1' ) then
4586      c          assign 20000 to err_return
4587      c
4588      c          type *, 'Please enter the number of the experiment to be restarted'
4589      c          type 90000, '          ((CR) for a list of applicable '
4590      c          1          // 'experiments): '
4591      c          accept 90010, answer
4592      c          read (unit=answer,fmt=*,err=70430) mi_pkey
4593      c          if ( mi_pkey .lt. 0 ) then
4594      c          70430      continue          ! Error in accepted
4595      c          mi_pkey = p_index( index_unit, read_unit1, read_unit2, '-X',
4596      c          1          .true. )
4597      c          if ( mi_pkey .eq. -1 ) then          ! No restartable exps.
4598      c          70000      continue
4599      c          type *
4600      c          type *, char( 7 ), 'Can't re-start.'
4601      c          type *
4602      c          go to 20000
4603      c          end if
4604      c          if ( mi_pkey .lt. 0 ) go to 20000 ! Operator cancelled.
4605      c          end if
4606      c
4607      c          70410      continue
4608      c          read (unit=index_unit,keyeq=mi_pkey,keyid=0,lostat=status)
4609      c          1          mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
4610      c          2          mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sir
4611      c          3          mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
4612      c          4          mi_tol, mi_which, mi_ofile, mi_anoises, mi_irlslm,
4613      c          5          mi_zeigs, mi_doas, mi_estype, mi_parp
4614      c          unlock (unit=index_unit)
4615      c          if ( status .eq. FOR$IOS_SPERECLLOC ) then
4616      c          type *, 'Record #', mi_pkey, ' locked. Waiting 1 sec.', uparrow
4617      c          call lib$wait( 1.0 )          ! Wait 1 second.
4618      c          type *, spaces, uparrow
4619      c          go to 70410
4620      c          else if ( ( status .ne. 0 )
4621      c          1          .or. ( index( '-X', mi_stat ) .le. 0 ) ) then
4622      c          go to 70000
4623      c          end if
4624      c
4625      c          havent_irec = .false.
4626      c          go to 51200
4627      c
4628      c!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!
4629      c
4630      c          New "trial" (rerun of initialized experiment) selected. Note
4631      c          that each "trial" will result in a NEW output file with the name
4632      c          6000*RT: ECHUCK.PARAMSJMUSIC_OUTPUTnnnnnnnnn.DAT, where nnnnnnnnn is

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4633 c      the (new) experiment number. Note also that, since each new "trial" !
4634 c      implies the generation of a new set of noise samples, this option !
4635 c      makes sense only for "ancestor" experiments. !
4636 c      !
4637 else if ( psel .eq. 'R' ) then
4638     assign 20000 to err_return
4639
4640 c      type *, 'Please enter the number of the experiment to be rerun'
4641     type 90000, '          (<CR> for a list of applicable '
4642     1                                           // 'experiments): '
4643     accept 90010, answer
4644     read (unit=answer,fmt=*,err=51430) mi_pkey
4645     if ( mi_pkey .lt. 0 ) then
4646 51430     continue ! Error in accepted #.
4647     mi_pkey = p_index( index_unit, read_unit1, read_unit2, ' IER',
4648     1                                           .true. )
4649     if ( mi_pkey .eq. -1 ) then ! No rerunnable experiments.
4650 51000     continue
4651     type *
4652     type *, char( 7 ), 'Can't re-run.'
4653     type *
4654     go to 20000
4655     end if
4656     if ( mi_pkey .lt. 0 ) go to 20000 ! Operator cancelled.
4657     end if
4658
4659 c      continue
4660 51410     read (unit=index_unit, keyeq=mi_pkey, keyid=0, iostat=status)
4661     1     mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
4662     2     mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
4663     3     mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
4664     4     mi_tol, mi_which, mi_ofile, mi_anoises, mi_irislim,
4665     5     mi_zeigs, mi_doas, mi_estype, mi_parp
4666     unlock (unit=index_unit)
4667     if ( status .eq. FOR$IOS_SPERECLOC ) then
4668     type *, 'Record #', mi_pkey, ' locked. Waiting 1 sec.', uparrow
4669     call lib$wait( 1.0 ) ! Wait 1 second.
4670     type *, spaces, uparrow
4671     go to 51410
4672     else if ( ( status .ne. 0 )
4673     1     .or. ( index( ' IER', mi_stat ) .le. 0 )
4674     2     .or. ( mi_park .ne. -1 )
4675     3     ) then
4676     go to 51000
4677     end if
4678
4679 c      Experiment to rerun has primary key value mi_pkey, and it is a
4680 c      valid rerun candidate. Now create required Virtual Arrays.
4681 c
4682     call init_va
4683     ants_by_2(1) = mi_ants
4684     ants_by_2(2) = 2
4685     waves_by_9(1) = mi_waves
4686     waves_by_9(2) = 9
4687     waves_by_2(1) = mi_waves
4688     waves_by_2(2) = 2
4689     nois_by_2(1) = mi_noises

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4690      nois_by_2(2)      = 2
4691      nois_by_3(1)      = mi_noises
4692      nois_by_3(2)      = 3
4693      ants_by_nois(1)   = mi_ants
4694      ants_by_nois(2)   = mi_noises
4695      if ( mi_which .gt. 0 ) then
4696          anois_by_2(1)  = mi_anoises
4697          anois_by_2(2)  = 2
4698          anois_by_3(1)  = mi_anoises
4699          anois_by_3(2)  = 3
4700          ants_by_anois(1) = mi_ants
4701          ants_by_anois(2) = mi_anoises
4702      end if
4703
4704      c
4705      call open_va( ant_add, ants_by_2, 2, dr_type, vstatus, msg )
4706      call exit_va( vstatus, msg )
4707      call open_va( wave_add, waves_by_9, 2, dr_type, vstatus, msg )
4708      call exit_va( vstatus, msg )
4709      call open_va( mtype_add, waves_by_2, 2, di_type, vstatus, msg )
4710      call exit_va( vstatus, msg )
4711      call open_va( ndef_add, nois_by_2, 2, dl_type, vstatus, msg )
4712      call exit_va( vstatus, msg )
4713      call open_va( npars_add, nois_by_3, 2, sr_type, vstatus, msg )
4714      call exit_va( vstatus, msg )
4715      call open_va( correl_add, ants_by_nois, 2, dr_type, vstatus, msg )
4716      call exit_va( vstatus, msg )
4717      if ( mi_which .gt. 0 ) then
4718          call open_va( andef_add, anois_by_2, 2, di_type, vstatus, msg )
4719          call exit_va( vstatus, msg )
4720          call open_va( anpars_add, anois_by_3, 2, sr_type, vstatus, msg )
4721          call exit_va( vstatus, msg )
4722          call open_va( acorrel_add, ants_by_anois, 2, dr_type, vstatus,
4723                      msg )
4724          call exit_va( vstatus, msg )
4725      end if
4726
4727      c
4728      c
4729      c
4730      c
4731      c
4732      c
4733      c
4734      c
4735      c
4736      c
4737      c
4738      c
4739      c
4740      c
4741      c
4742      c
4743      c
4744      c
4745      c
4746      c

```

Get problem definition from input file, initialize the noise counters, and form the array phase shift matrix.

```

work_file = 'MUSIC_INPUT_____DAT'
write (unit=work_file( 12 : 21 ),fmt=90020,lostat=status,
1      err=60510) mi_pkey
c
open (unit=data_unit,file='6000*RT:[CHUCK.PARAMS]//work_file,
1      status='OLD',organization='SEQUENTIAL',access='SEQUENTIAL',
2      form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
3      READONLY,lostat=status,err=60560)
read (unit=data_unit,err=60500)
if ( .not. (
1      readarr( %val( ant_add ), dr_type, mi_ants, 2, data_unit, 32 )
2      ) ) go to 60500
if ( .not. (
1      readarr( %val( wave_add ), dr_type, mi_waves, 9, data_unit,
2      32 )
3      ) ) go to 60500
if ( .not. (
1      readarr( %val( mtype_add ), di_type, mi_waves, 2, data_unit,

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4747          2          64 )
4748          3          ) go to 60500
4749      if ( .not. (
4750          1      readarr( %val( ndef_add ), di_type, mi_noises, 2, data_unit,
4751          2          64 )
4752          3          ) ) go to 60500
4753      if ( .not. (
4754          1      readarr( %val( npars_add ), sr_type, mi_noises, 3, data_unit,
4755          2          64 )
4756          3          ) ) go to 60500
4757      if ( .not. (
4758          1      readarr( %val( correl_add ), dr_type, mi_ants, mi_noises,
4759          2          data_unit, 32 )
4760          3          ) ) go to 60500
4761      if ( mi_which .gt. 0 ) then
4762          if ( .not. (
4763          1      readarr( %val( andef_add ), di_type, mi_anoises, 2,
4764          2          data_unit, 64 )
4765          3          ) ) go to 60500
4766          if ( .not. (
4767          1      readarr( %val( anpars_add ), sr_type, mi_anoises, 3,
4768          2          data_unit, 64 )
4769          3          ) ) go to 60500
4770          if ( .not. (
4771          1      readarr( %val( acorrel_add ), dr_type, mi_ants, mi_anoises,
4772          2          data_unit, 32 )
4773          3          ) ) go to 60500
4774      end if
4775      close (unit=data_unit)
4776      c
4777      havent_irec = .true.
4778      go to 51210
4779      c
4780      c!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!
4781      c
4782      c      New experiment selected.
4783      c
4784      else if ( psel .eq. 'N' ) then
4785      assign 20000 to err_return
4786      havent_irec = .true.
4787      c
4788      51200      continue          ! Branch to restart a failed experiment.
4789      type *,
4790      type *, '      Take defaults from which experiment number?'
4791      if ( havent_irec ) then
4792      type *, '          (-1 for "standard" defaults)'
4793      else
4794      type *, '          (-1 for current values)'
4795      end if
4796      type 90000, ' (<CR> to see a list of available experiments): '
4797      accept 90010, answer
4798      read (unit=answer,fmt=*,err=51100) selexp
4799      go to 51110
4800      c
4801      51100      continue          ! Error in accepted #.
4802      selexp = p_index( index_unit, read_unit1, read_unit2,
4803      1          'IEDR', .true. )

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4804          if ( selexp .lt. -1 ) go to 20000          ! Operator cancelled.
4805      c
4806      51110      continue
4807          if ( selexp .ge. 0 ) then                    ! Defaults specified.
4808      51120      continue
4809          read (unit=index_unit,keyeq=selexp,keyid=0,iostat=status)
4810      1          selexp, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
4811      2          mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
4812      3          mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
4813      4          mi_tol, mi_which, mi_ofile, mi_anoises, mi_irlslim,
4814      5          mi_zeigs, mi_doas, mi_estype, mi_parp
4815          unlock (unit=index_unit)
4816          if ( status .eq. FOR$IOS_SPERECLOC ) then
4817              type *, 'Record #', selexp, ' locked. Waiting 1 sec.', uparrow
4818              call lib$wait( 1.0 )                    ! Wait 1 second.
4819              type *, spaces, uparrow
4820              go to 51120
4821          else if ( ( status .ne. 0 )
4822      1          .or. ( index( 'IEDR', mi_stat ) .le. 0 ) ) then
4823      51150      continue
4824              type *
4825              type *, char( 7 ), 'Can't use that experiment for defaults.'
4826              type *, 'Setting "standard" defaults.'
4827              type *
4828              selexp = -1
4829              go to 51130
4830          end if
4831      c
4832      c          Experiment from which to obtain default values has primary key
4833      c          value selexp, and it is a valid defaults candidate. Replace it with
4834      c          its "ancestor" if it is a "descendant" experiment.
4835      c
4836          do while ( mi_park .ne. -1 )                  ! "Descendant".
4837          selexp = mi_park
4838      51140      continue
4839          read (unit=index_unit,keyeq=selexp,keyid=0,iostat=status)
4840      1          selexp, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
4841      2          mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
4842      3          mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
4843      4          mi_tol, mi_which, mi_ofile, mi_anoises, mi_irlslim,
4844      5          mi_zeigs, mi_doas, mi_estype, mi_parp
4845          unlock (unit=index_unit)
4846          if ( status .eq. FOR$IOS_SPERECLOC ) then
4847              type *, 'Record #', selexp, ' locked. Waiting 1 sec.', uparrow
4848              call lib$wait( 1.0 )                    ! Wait 1 second.
4849              type *, spaces, uparrow
4850              go to 51140
4851          else if ( status .ne. 0 ) then                ! Error reading index.
4852              go to 51150
4853          end if
4854          end do
4855      c
4856          else if ( havent_irec ) then                  ! No defaults spec'ed.
4857              mi_park = -1
4858              mi_ants = 5
4859              mi_waves = 3
4860              mi_noises = 5

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4861         mi_smp = 101
4862         mi_sint = .25
4863         mi_begp = 1.5
4864         mi_endp = 2.0
4865         mi_delp = 0.5
4866         mi_peps = 0.01
4867         mi_norm = 1.0
4868         mi_wlen = 3.0
4869         mi_tol = 1.0e-10
4870         mi_which = 1
4871         mi_anoises = 6
4872         mi_irlslim = 100
4873         mi_zeigs = 0
4874         mi_doas = 0
4875         mi_estype = 0
4876         mi_parp = 0.0
4877     end if
4878 c
4879 51130     continue
4880     call date( mi_dtim( 1:9 ) )
4881     mi_dtim( 10:11 ) = ' '
4882     call time( mi_dtim( 12:19 ) )
4883 c
4884     old_ants = mi_ants
4885     old_waves = mi_waves
4886     old_noises = mi_noises
4887     old_anoises = mi_anoises
4888     old_which = mi_which
4889     old_smp = mi_smp
4890     call get_music_def( mi_ants, mi_waves, mi_noises,
4891 1         mi_anoises, mi_which, mi_smp )
4892 c
4893 c         Now create required Virtual Arrays.
4894 c
4895     call init_va
4896     ants_by_2(1) = mi_ants
4897     ants_by_2(2) = 2
4898     waves_by_9(1) = mi_waves
4899     waves_by_9(2) = 9
4900     waves_by_2(1) = mi_waves
4901     waves_by_2(2) = 2
4902     nois_by_2(1) = mi_noises
4903     nois_by_2(2) = 2
4904     nois_by_3(1) = mi_noises
4905     nois_by_3(2) = 3
4906     ants_by_nois(1) = mi_ants
4907     ants_by_nois(2) = mi_noises
4908     if ( mi_which .gt. 0 ) then
4909         anois_by_2(1) = mi_anoises
4910         anois_by_2(2) = 2
4911         anois_by_3(1) = mi_anoises
4912         anois_by_3(2) = 3
4913         ants_by_anois(1) = mi_ants
4914         ants_by_anois(2) = mi_anoises
4915     end if
4916 c
4917     call open_va( ant_add, ants_by_2, 2, dr_type, vstatus, msg )

```

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4918      call exit_va( vstatus, msg )
4919      call open_va( wave_add, waves_by_9, 2, dr_type, vstatus, msg )
4920      call exit_va( vstatus, msg )
4921      call open_va( mtype_add, waves_by_2, 2, di_type, vstatus, msg )
4922      call exit_va( vstatus, msg )
4923      call open_va( ndef_add, nois_by_2, 2, di_type, vstatus, msg )
4924      call exit_va( vstatus, msg )
4925      call open_va( npars_add, nois_by_3, 2, sr_type, vstatus, msg )
4926      call exit_va( vstatus, msg )
4927      call open_va( correl_add, ants_by_nois, 2, dr_type, vstatus, msg )
4928      call exit_va( vstatus, msg )
4929      if ( mi_which .gt. 0 ) then
4930          call open_va( andef_add, anois_by_2, 2, di_type, vstatus, msg )
4931          call exit_va( vstatus, msg )
4932          call open_va( anpars_add, anois_by_3, 2, sr_type, vstatus, msg )
4933          call exit_va( vstatus, msg )
4934          call open_va( acorrel_add, ants_by_anois, 2, dr_type, vstatus,
4935      1          msg )
4936      call exit_va( vstatus, msg )
4937      end if
4938      c
4939      c
4940      c      Get antenna positions, wave characteristics, noise definitions,
4941      c      sample interval, tolerance, and parameters for Lp estimate calculation.
4942      c      First get values from selected default experiment, if any.
4943      c
4944      if ( selexp .ge. 0 ) then
4945          ! If default was selected.
4946          work_file = 'MUSIC_INPUT_____DAT'
4947          write (unit=work_file( 12 : 21 ),fmt=90020,iostat=status,
4948      1          err=51310) selexp
4949          open (unit=data_unit,file='6000$RT:[CHUCK.PARAMS]//work_file,
4950      1          status='OLD',organization='SEQUENTIAL',access='SEQUENTIAL',
4951      2          form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
4952      3          READONLY,iostat=status,err=51310)
4953          read (unit=data_unit,err=51300)
4954          if ( old_ants .eq. mi_ants ) then
4955              if ( .not. ( readarr( %val( ant_add ), dr_type,
4956      1          mi_ants, 2, data_unit, 32 ) ) ) go to 51300
4957          else
4958              if ( .not. ( skiparr( old_ants, 2, data_unit, 32 ) ) )
4959      1          ) go to 51300
4960          end if
4961          if ( old_waves .eq. mi_waves ) then
4962              if ( .not. ( readarr( %val( wave_add ), dr_type,
4963      1          mi_waves, 9, data_unit, 32 ) ) ) go to 51300
4964          else
4965              if ( .not. ( readarr( %val( mtype_add ), di_type,
4966      1          mi_waves, 2, data_unit, 64 ) ) ) go to 51300
4967          else
4968              if ( .not. ( skiparr( old_waves, 9, data_unit, 32 ) ) )
4969      1          ) go to 51300
4970              if ( .not. ( skiparr( old_waves, 2, data_unit, 64 ) ) )
4971      1          ) go to 51300
4972          end if
4973          if ( old_noises .eq. mi_noises ) then
4974              if ( .not. ( readarr( %val( ndef_add ), di_type,
4975      1          mi_noises, 2, data_unit, 64 ) ) ) go to 51300
4976          else
4977              if ( .not. ( readarr( %val( npars_add ), sr_type,
4978      1          mi_noises, 3, data_unit, 64 ) ) ) go to 51300

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4975         if ( old_ants .eq. mi_ants ) then
4976             if ( .not. ( readarr( %val( correl_add ), dr_type, mi_ants,
4977                 1             mi_noises, data_unit, 32 ) ) ) go to 51300
4978         else
4979             if ( .not. ( skiparr( old_ants, old_noises, data_unit, 64 ) )
4980                 ) go to 51300
4981         end if
4982     else
4983         if ( .not. ( skiparr( mi_noises, 2, data_unit, 64 ) )
4984             ) go to 51300
4985         if ( .not. ( skiparr( mi_noises, 3, data_unit, 64 ) )
4986             ) go to 51300
4987         if ( .not. ( readarr( %val( correl_add ), dr_type, mi_ants,
4988             1             mi_noises, data_unit, 32 ) ) ) go to 51300
4989     end if
4990     if ( ( mi_which .gt. 0 )
4991         .and. ( old_which .gt. 0 )
4992         .and. ( old_anoise .eq. mi_anoise ) ) then
4993         if ( .not. ( readarr( %val( andef_add ), di_type,
4994             1             mi_anoise, 2, data_unit, 64 ) ) ) go to 51300
4995         if ( .not. ( readarr( %val( anpars_add ), sr_type,
4996             1             mi_anoise, 3, data_unit, 64 ) ) ) go to 51300
4997         if ( old_ants .eq. mi_ants ) then
4998             if ( .not. ( readarr( %val( acorrel_add ), dr_type, mi_ants,
4999                 1             mi_anoise, data_unit, 32 ) ) ) go to 51300
5000         end if
5001     end if
5002     close (unit=data_unit)
5003 end if
5004 go to 51320
5005 c
5006 c             Error reading defaults branch - abandon defaults.
5007 c
5008 51300 continue
5009 close (unit=data_unit)
5010 c
5011 51310 continue
5012 selexp = -1
5013 type *
5014 type *, 'Error accessing defaults definition file.'
5015 type *, 'Setting "standard" defaults.'
5016 type *
5017 c
5018 51320 continue
5019 call read_antennae( %val( ant_add ), mi_ants, old_ants, selexp )
5020 call read_waves( mi_waves, old_waves, selexp, %val( wave_add ),
5021     1             %val( mtype_add ) )
5022 if ( mi_which .gt. 0 ) then             ! Get actual noise?
5023     call read_noise( selexp, mi_ants, old_ants, mi_anoise,
5024         1             old_anoise, %val( andef_add ), %val( anpars_add ),
5025         2             %val( acorrel_add ), -1, old_which )
5026 end if
5027 call read_noise( selexp, mi_ants, old_ants, mi_noises,
5028     1             old_noises, %val( ndef_add ), %val( npars_add ),
5029     2             %val( correl_add ), mi_which, old_which )
5030 if ( havent_irec ) then             ! New, not recreate.
5031     call read_misc( selexp, mi_wlen, mi_sint, mi_tol )

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5032          call read_lp( selexp, mi_begp, mi_endp, mi_delp,
5033          1          mi_peps, mi_norm, mi_irislim )
5034          end if
5035      c
5036      c          Convert input scalar wavelength to vector of (identical)
5037      c          wavelengths.
5038      c
5039      c          call dr_colfill( %val( wave_add ), mi_waves, 9, mi_wlen, 9 )
5040      c
5041      c          Obtain an (unused) experiment number, create a (hopefully)
5042      c          unique output file name, and enter the initial experiment description
5043      c          in the index.
5044      c
5045      51210      continue          ! Branch to create new trial or experiment.
5046      c          if ( havent_irec ) then          ! New, not recreate.
5047      c          mi_pkey = avlbl_prikey( index_unit, 0 )
5048      c          if ( mi_pkey .lt. 0 ) go to 50000
5049      c
5050      c          mi_stat = '--'
5051      c          mi_ierr = 0
5052      c          mi_eerr = 0
5053      c          mi_serr = 0
5054      c
5055      c          mi_ofile = '6000*RT:[CHUCK.PARAMS]MUSIC_OUTPUT_____DAT'
5056      c          write (unit=mi_ofile( 35 : 44 ),fmt=90020,err=60510) mi_pkey
5057      c
5058      c          write (unit=index_unit,err=50000)
5059      1          mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
5060      2          mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
5061      3          mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
5062      4          mi_tol, mi_which, mi_ofile, mi_anoises, mi_irislim,
5063      5          mi_zeigs, mi_doas, mi_estype, mi_parp
5064      c          unlock (unit=index_unit)
5065      c          end if
5066      c
5067      c          work_file = 'MUSIC_INPUT_____DAT'
5068      c          write (unit=work_file( 12 : 21 ),fmt=90020,iostat=status,
5069      1          err=60510) mi_pkey
5070      c
5071      c          call lib$delete_file( '6000*RT:[CHUCK.PARAMS]' // work_file // '.*' )
5072      c          open (unit=data_unit,file='6000*RT:[CHUCK.PARAMS]//work_file,
5073      1          status='NEW',organization='SEQUENTIAL',access='SEQUENTIAL',
5074      2          form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
5075      3          iostat=status,err=60510)
5076      c          write (unit=data_unit,err=60500)
5077      1          mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
5078      2          mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
5079      3          mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
5080      4          mi_tol, mi_which, mi_ofile, mi_anoises, mi_irislim,
5081      5          mi_zeigs, mi_doas, mi_estype, mi_parp
5082      c          if ( .not. ( writearr( %val( ant_add ), dr_type,
5083      1          mi_ants, 2, data_unit, 32 ) ) ) go to 60500
5084      c          if ( .not. ( writearr( %val( wave_add ), dr_type,
5085      1          mi_waves, 9, data_unit, 32 ) ) ) go to 60500
5086      c          if ( .not. ( writearr( %val( mtype_add ), di_type,
5087      1          mi_waves, 2, data_unit, 64 ) ) ) go to 60500
5088      c          if ( .not. ( writearr( %val( ndef_add ), di_type,

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5089         mi_noises, 2, data_unit, 64 ) ) go to 60500
5090     if ( .not. ( writearr( %val( npars_add ), sr_type,
5091         mi_noises, 3, data_unit, 64 ) ) ) go to 60500
5092     if ( .not. ( writearr( %val( correl_add ), dr_type, mi_ants,
5093         mi_noises, data_unit, 32 ) ) ) go to 60500
5094     if ( mi_which .gt. 0 ) then
5095         if ( .not. ( writearr( %val( andef_add ), di_type,
5096         mi_anaises, 2, data_unit, 64 ) ) ) go to 60500
5097         if ( .not. ( writearr( %val( anpars_add ), sr_type,
5098         mi_anaises, 3, data_unit, 64 ) ) ) go to 60500
5099         if ( .not. ( writearr( %val( acorrel_add ), dr_type, mi_ants,
5100         mi_anaises, data_unit, 32 ) ) ) go to 60500
5101     end if
5102     close (unit=data_unit)
5103     c
5104     c           Update index file entry to reflect creation of the input file.
5105     c
5106     mi_stat = ' '
5107     go to 60600
5108     c
5109     c           Special file I/O error path for creating a new experiment.
5110     c
5111     50000   continue
5112           call dump_va( vstatus, msg )
5113           call exit_va( vstatus, msg )
5114           type *
5115           type *, 'Error initializing experiment; new trial / experiment ',
5116           'not created.'
5117           type *, '(And you can't do a thing about it!)'
5118           type *
5119           go to 20000
5120     c
5121     c!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!
5122     c
5123     c           Exit to operating system selected.
5124     c
5125     c           else if ( psel .eq. 'X' ) then
5126           status = sys$setprn( procname( 1 : pname_len ) )
5127           if ( .not. status ) call lib$stop( %val( status ) )
5128           stop
5129     c
5130     c!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!
5131     c
5132     c           No legitimate selection made.
5133     c
5134     end if
5135     go to 20000
5136     c
5137     c!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!
5138     c*****
5139     c
5140     c           Branch to rewrite the currently selected experiment's index
5141     c           record with a revised status (mi_stat set to desired new value). If
5142     c           the rewrite is successful, this program segment returns to statement
5143     c           number 20000, otherwise index-record I/O error processing is initiated.
5144     c
5145     60600   continue

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5146      read (unit=index_unit,key=mi_pkey,keyid=0,iostat=status)
5147      if ( status .eq. FOR$IOS_SPERECLD ) then
5148          unlock (unit=index_unit)
5149          type *, 'Index record locked by another program ...'
5150          call lib$wait( 10.0 ) ! Wait 10 seconds.
5151          type *, 'Trying again.'
5152          go to 60600
5153      else if ( status .ne. 0 ) then
5154          unlock (unit=index_unit)
5155          go to 60700
5156      end if
5157      rewrite (unit=index_unit,iostat=status,err=60700)
5158      1          mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
5159      2          mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
5160      3          mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
5161      4          mi_totl, mi_which, mi_ofile, mi_anosies, mi_lrlslim,
5162      5          mi_zeigs, mi_doas, mi_estype, mi_parp
5163      unlock (unit=index_unit)
5164      go to 20000
5165      c
5166      c*****
5167      c
5168      c          File error processing branches.
5169      c
5170      c
5171      c          Error opening intermediate data file - could be locked.
5172      c
5173      60550      continue
5174      close (unit=data_unit)
5175      c
5176      60560      continue
5177      call dump_va( vstatus, msg )
5178      call exit_va( vstatus, msg )
5179      c
5180      60570      continue
5181      type *
5182      type *, char( 7 ), 'Can''t open required data file.'
5183      type *, 'File may be in use by another program.'
5184      type *
5185      go to err_return
5186      c
5187      c          Error reading an open intermediate data file - must be some
5188      c          problem with the file itself.
5189      c
5190      60500      continue
5191      close (unit=data_unit)
5192      c
5193      60510      continue
5194      call dump_va( vstatus, msg )
5195      call exit_va( vstatus, msg )
5196      c
5197      60540      continue
5198      type *
5199      type *, char( 7 ), 'Error finding or accessing data file(s).'
5200      type 90030, char( 7 ) // 'Experiment #', mi_pkey, ' '
5201      type 90040, 'FORTRAN error code #', status, ' (hex).'
5202      type *, char( 7 ), 'File: ', work_file

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5203      type *
5204      type 90000, ' Mark ALL intermediate files for that experiment '
5205      1 // 'for DELETION (Y/[N])? '
5206      accept 90010, answer
5207      psel = cap( answer( 1 : 1 ) )
5208  c
5209      if ( psel .eq. 'Y' ) then
5210          type 90000, ' ' // char( 7 ) // 'Are you sure (Y/[N])? '
5211          accept 90010, answer
5212          psel = cap( answer( 1 : 1 ) )
5213          if ( psel .ne. 'Y' ) go to 60520
5214  c
5215      60530      continue
5216      read (unit=index_unit,key=mi_pkey,keyld=0,lostat=status)
5217      if ( status .eq. FOR$IOS_SPERECLC ) then
5218          unlock (unit=index_unit)
5219          type *, 'Index record locked by another program ...'
5220          call lib$wait( 10.0 ) ! Wait 10 seconds.
5221          type *, 'Trying again.'
5222          go to 60530
5223      else if ( status .ne. 0 ) then
5224          unlock (unit=index_unit)
5225          go to 60700
5226      end if
5227      mi_stat = 'X'
5228      rewrite (unit=index_unit,lostat=status,err=60700)
5229      1      mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
5230      2      mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
5231      3      mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
5232      4      mi_tol, mi_which, mi_ofile, mi_anoises, mi_irislim,
5233      5      mi_zeigs, mi_doas, mi_estype, mi_parp
5234      unlock (unit=index_unit)
5235  c
5236      type *
5237      type 90030, char( 7 ) // 'Experiment #', mi_pkey, ' marked for '
5238      1 // 'DELETION!'
5239      type *
5240      else
5241      60520      continue
5242      type *
5243      type 90030, 'Index record for experiment #', mi_pkey, ' not changed.'
5244      type *
5245      end if
5246      go to err_return
5247  c
5248      Failed access to index record - not a lot we can do about it.
5249  c
5250      60700      continue
5251      type *
5252      type *, char( 7 ), 'Read or write failure accessing index record!'
5253      type 90030, char( 7 ) // 'Experiment #', mi_pkey, '
5254      type 90040, 'FORTRAN error code #', status, ' (hex).'
5255      type *, char( 7 ), 'Status of that index record is uncertain.'
5256      type *
5257      go to 20000
5258  c
5259      c*****

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5260      c
5261      90000      format( $a )
5262      90010      format( a10 )
5263      90020      format( i10.10 )
5264      90030      format( 1x, a, i10.10, a )
5265      90040      format( 1x, a, z8.8, a )
5266      90050      format( $, 1x, 2( a, i6 ), a )
5267      c
5268      end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	25984	PIC CON REL LCL SHR EXE RD NOWRT LC
1 \$PDATA	5492	PIC CON REL LCL SHR NOEXE RD NOWRT LC
2 \$LOCAL	6308	PIC CON REL LCL NOSHR NOEXE RD WRT QL
3 MUSIC_MEASURES	120	PIC OVR REL GBL SHR NOEXE RD WRT LC
Total Space Allocated		37904

ENTRY POINTS

Address	Type	Name
0-00000000		DO_MUSIC

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type	Name
2-00000388	I*4	ACORREL_ADD	2-00000318	L*4	ANCESTOR	2-00000380	I*4	ANDEF
2-000001A0	CHAR	ANSWER	2-00000358	I*4	ANT_ADD	2-0000019D	CHAR	CHSEL
2-0000036C	I*4	CORREL_ADD	2-000003A0	I*4	COUNT	3-0000000C	I*4	CYCLE
2-00000354	I*4	DATA_UNIT2	2-00000308	L*4	DISP_INDEX	2-00000394	I*4	DOA_f
2-00000390	I*4	EIGV_ADD	2-0000038C	I*4	EIG_ADD	2-000003C0	I*4	ELSEL
2-000003D0	I*4	ESTYPE	2-00000398	I*4	EXNM_ADD	2-000003A4	I*4	FSKIF
2-0000039C	I*4	I	2-00000344	I*4	INDEX_UNIT	2-000003A8	I*4	ISKIF
2-000003F4	I*4	MI_ANDISES	2-000003E0	I*4	MI_ANTS	2-00000408	R*4	MI_BE
2-00000400	I*4	MI_DOAS	2-00000278	CHAR	MI_DTIM	2-000002D8	I*2	MI_EE
2-00000404	I*4	MI_ESTYPE	2-000002D6	I*2	MI_IERR	2-000003F8	I*4	MI_IR
2-000002F0	R*8	MI_NORM	2-00000286	CHAR	MI_OFIL	2-000003DC	I*4	MI_PA
2-000002E8	R*8	MI_PEPS	2-000003D8	I*4	MI_PKEY	2-000002DA	I*2	MI_SE
2-000003EC	I*4	MI_SMP	2-00000272	CHAR	MI_STAT	2-00000300	R*8	MI_TC
2-000003F0	I*4	MI_WHICH	2-000002F8	R*8	MI_WLEN	2-000003FC	I*4	MI_ZE
2-00000360	I*4	MTYPE_ADD	2-00000364	I*4	NDEF_ADD	2-00000374	I*4	NOISE
3-00000068	I*4	NON_TRI_CYC	2-00000368	I*4	NPARS_ADD	2-000003B4	I*4	NUM_A
2-000003B8	I*4	NUM_EXPS	2-000003AC	I*4	NUM_ZEIGS	2-00000370	I*4	OBSMA
2-0000032C	I*4	OLD_ANDISES	2-00000320	I*4	OLD_ANTS	2-00000328	I*4	OLD_N
2-00000324	I*4	OLD_WAVES	2-00000330	I*4	OLD_WHICH	2-000003D4	R*4	P
2-0000033C	I*4	PNAME_LEN	2-000001AA	CHAR	PROCNAME	2-00000198	CHAR	PSEL
3-0000006C	R*8	QPS1_LMARG	3-00000074	I*4	QPS1_L_ZCYC	2-00000348	I*4	READ_

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```

0001 c
0002 c      integer*4 function p_index( index_unit, read_unit1, read_unit2,
0003 c      1                                     keyrange, show )
0004 c
0005 c=====
0006 c
0007 c      This function finds and (optionally) displays records from
0008 c      MUSIC experiments index file open on logical unit numbers index_u
0009 c      read_unit1, and read_unit2 (ALL LUNs attached to the SAME file).
0010 c      parameters index_unit, read_unit1, and read_unit2 are INTEGER*4,
0011 c      keyrange is CHARACTER*(*), and show is LOGICAL.
0012 c      If show is .true., the records to be displayed are select
0013 c      keyrange as follows: If keyrange is '*', all records from the se
0014 c      MUSIC index file are displayed. Otherwise, the records whose
0015 c      "statuses" (first alternate key field) match the characters in th
0016 c      string keyrange are displayed, in the order of the characters in
0017 c      keyrange. For example, if keyrange is 'SR', all 'S' records are
0018 c      displayed, followed by all 'R' records.
0019 c      If show is .true., the function returns the number the op
0020 c      enters in response to a MUSIC_INDEX_DISPLAY routine prompt, or -1
0021 c      matching records exist (no records at all if keyrange = '*'), or
0022 c      one or more records are displayed but the operator makes no recor
0023 c      selection.
0024 c      If show is .false., the function simply tests for the pre
0025 c      of matching records (any records if keyrange = '*') in the select
0026 c      MUSIC index file. If any matching records are found, the functio
0027 c      returns 0, otherwise it returns -1.
0028 c      The three LUNs index_unit, read_unit1, and read_unit2 are
0029 c      handled as follows: If keyrange = '*', read_unit1 and read_unit2
0030 c      used to read the file in SECOND ALTERNATE / PRIMARY key HEIRARCHY
0031 c      order; otherwise read_unit1 is used to read the file in FIRST ALT:
0032 c      key ASCENDING order. The LUN index_unit is used by routine
0033 c      MUSIC_INDEX_DISPLAY (if show is .true.), to read the file in PRIM
0034 c      key order only.
0035 c
0036 c=====
0037 c
0038 c      implicit none
0039 c      intrinsic len, index, char
0040 c      external next_rec, music_index_display
0041 c
0042 c      logical next_rec
0043 c      integer*4 music_index_display
0044 c
0045 c      logical show
0046 c      integer*4 index_unit, read_unit1, read_unit2
0047 c      character*(*) keyrange
0048 c
0049 c      logical notall, some_found, tlog
0050 c      integer*4 ikey, status, slen, chno, iskey
0051 c      character*1 ch
0052 c
0053 c      some_found = .false.          ! No records found yet.
0054 c      tlog = next_rec( read_unit1, read_unit2, ikey, char( 0 ), iskey )
0055 c      notall = index( keyrange, '*' ) .le. 0 ! Not all statuses select
0056 c      if ( notall ) then
0057 c          chno = 1

```

```

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0058          slen = len( keyrange )
0059      else
0060          ch = '*'
0061      end if
0062      c
0063      10000  continue
0064          if ( notall ) ch = keyrange( chno : chno )
0065      c
0066      10010  continue
0067          if ( .not. next_rec( read_unit1, read_unit2, ikey, ch, iskey ) )
0068      1                                           go to 10500
0069      c
0070      10100  continue
0071          if ( show ) then
0072      10130  continue
0073          p_index = music_index_display( index_unit, ikey, .true. )
0074          some_found = .true.
0075          if ( p_index .eq. -1 ) then                ! Display not full.
0076      10110  continue
0077          if ( .not. next_rec( read_unit1, read_unit2, ikey, ch, iskey ) )
0078      1                                           go to 10200
0079          go to 10130
0080          else if ( p_index .eq. -2 ) then          ! User wants next pg.
0081      10120  continue
0082          if ( .not. next_rec( read_unit1, read_unit2, ikey, ch, iskey ) )
0083      1                                           go to 10300
0084          go to 10130
0085          else                                     ! User entered a number (since <CR> on non-
0086      c                                           existent record is handled by status
0087      c                                           checks).
0088      c
0089          tlog = next_rec( read_unit1, read_unit2, ikey, char( 0 ), iskey )
0090          return
0091      end if
0092      else                                         ! Find only, a matching record found.
0093          p_index = 0
0094          tlog = next_rec( read_unit1, read_unit2, ikey, char( 0 ), iskey )
0095          return
0096      end if
0097      c
0098      10200  continue
0099      c
0100          End of selected index records, ( maximum records currently
0101      c      displayed.
0102      c
0103          if ( notall .and. ( chno .lt. slen ) ) then
0104      c
0105          Multiple secondary key values specified, and values remain to
0106      c      be processed.
0107      c
0108          chno = chno + 1
0109          go to 10000
0110          end if
0111      c
0112          All secondary key values processed, or primary key search was
0113      c      performed, and display was selected.
0114      c

```

```

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0115      p_index = music_index_display( index_unit, -1, .true. )
0116      if ( p_index .lt. 0 ) p_index = -2 ! Indicate operator cancelled.
0117      tlog = next_rec( read_unit1, read_unit2, ikey, char( 0 ), iskey )
0118      return
0119      c
0120      10300      continue
0121      c
0122      End of selected index records, with maximum currently
0123      displayed, operator has solicited the next page, and display was
0124      selected.
0125      c
0126      if ( notall .and. ( chno .lt. slen ) ) then
0127      c
0128      Multiple secondary key values specified, and values remain to
0129      be processed.
0130      c
0131      chno = chno + 1
0132      go to 10000
0133      end if
0134      c
0135      All secondary key values processed, or primary key search was
0136      performed, and display was selected.
0137      c
0138      p_index = -2 ! Indicate operator cancelled.
0139      tlog = next_rec( read_unit1, read_unit2, ikey, char( 0 ), iskey )
0140      return
0141      c
0142      Error on read of initial record with the current selected key.
0143      c
0144      10500      continue
0145      if ( notall .and. ( chno .lt. slen ) ) then
0146      c
0147      Multiple secondary key values specified, and values remain to
0148      be processed.
0149      c
0150      chno = chno + 1
0151      go to 10000
0152      end if
0153      c
0154      All secondary key values processed, or primary key search was
0155      performed.
0156      c
0157      if ( some_found .and. show ) then
0158      p_index = music_index_display( index_unit, -1, .true. )
0159      if( p_index .lt. 0 ) p_index = -2 ! Indicate operator cancelled.
0160      else
0161      p_index = -1 ! Indicate nothing available.
0162      end if
0163      tlog = next_rec( read_unit1, read_unit2, ikey, char( 0 ), iskey )
0164      return
0165      c
0166      end

```

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```

0001 c
0002 c      logical function next_rec( unit1, unit2, pkey, skey1, skey2 )
0003 c
0004 c=====
0005 c
0006 c      This function determines the key values for the next MUSIC
0007 c      index record meeting a search criterion. The MUSIC index file to be
0008 c      searched is assumed to be open on Logical Unit Numbers unit1 and unit2
0009 c      (BOTH LUNs open on the SAME file).
0010 c      The search criterion to be used depends on the CHARACTER*1
0011 c      value skey1 on entry. If skey1 is not '*', records whose first
0012 c      alternate key values ("statuses") are equal to skey1 are sought. If
0013 c      skey1 = '*', records are sought in "heirarchical" order. When
0014 c      "heirarchical" order is used, an "ancestor" record is returned first,
0015 c      followed by its "descendant" records, followed by the next "ancestor"
0016 c      record, and so on. "Descendant" records are presented in a similar
0017 c      order; i.e. each record is followed directly by its "descendant"
0018 c      records, followed by the next record which is at the same level of
0019 c      "descendancy" as the first "descendant" record.
0020 c      Since this routine is designed to be called repeatedly to
0021 c      perform a particular search, a reset MUST BE PERFORMED when the desired
0022 c      search has been completed. A reset of this routine is accomplished by
0023 c      invoking the function with skey1 set to ASCII NULL (ASCII value of 0).
0024 c      The parameters unit1, unit2, pkey, and skey2 are INTEGER*4,
0025 c      while skey1 is CHARACTER*1. The primary key value of the next
0026 c      applicable record (if any) is returned in pkey, and its second
0027 c      alternate key value (its "parent" primary key value) is returned in
0028 c      skey2. NOTE that the parameter skey1 is NOT written (value NOT
0029 c      changed).
0030 c      The function return value is .true. if a matching record was
0031 c      found, and .false. if not. It is also set to .true. when a reset is
0032 c      performed.
0033 c
0034 c=====
0035 c
0036 c      implicit none
0037 c      intrinsic ichar
0038 c      external lib$get_lun, lib$stop, lib$free_lun, lib$wait
0039 c      include 'SYS$LIBRARY:FOR IOSDEF'
0040 c
0041 c      Integer*4 lib$get_lun
0042 c
0043 c      Integer*4 unit1, unit2, pkey, skey2
0044 c      character*1 skey1
0045 c
0046 c      Integer*4 sort_unit, status, sort_open/ .false. /, stage/ -1 /
0047 c      integer*4 old_sk1/ -1 /, rd_ptr, ls_ptr, ins_ptr, shf_ptr
0048 c      integer*4 tpkey, tskey, parkey
0049 c      character*1 tch
0050 c      character*1 uparrow( 3 )/ 27, 'C', 'A' /, spaces( 70 )/ 70*' ' /
0051 c
0052 c      save sort_unit, sort_open, stage, old_sk1, rd_ptr, ls_ptr, ins_ptr
0053 c
0054 c      if ( ichar( skey1 ) .eq. 0 ) then          ! Reset to be performed.
0055 c          if ( sort_open ) then
0056 c              close (unit=sort_unit)
0057 c              call lib$free_lun( sort_unit )

```



```

NEXT_REC                                18-Apr-1988 14:00:27    VAX FI
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0143          sort_open = .false.
0144          end if
0145          old_sk1 = -1
0146          stage = -1
0147          next_rec = .true.
0148          return                                ! Reset completed.
0149      end if
0150  c
0151      if ( stage .eq. -1 ) then
0152          if ( ( skey1 .eq. '*' ) .and. ( .not. sort_open ) ) then
0153              status = lib$get_lun( sort_unit )
0154              if ( .not. status ) call lib$stop( %val( status ) )
0155              open (unit=sort_unit,file='6000*RT:[CHUCK.PARAMS]SORT_FILE.DAT',
0156                  1 form='UNFORMATTED',recordtype='FIXED',recl=2,
0157                  2 organization='RELATIVE',access='DIRECT',status='UNKNOWN',
0158                  3 dispose='DELETE')
0159              sort_open = .true.
0160          end if
0161          stage = 0
0162      end if
0163  c
0164      if ( ( skey1 .ne. '*' ) .and. ( ichar( skey1 ) .ne. old_sk1 ) ) then
0165          old_sk1 = ichar( skey1 )
0166          stage = 0
0167      end if
0168  c
0169      if ( stage .eq. 0 ) then
0170          10000 continue
0171          if ( skey1 .eq. '*' ) then
0172              read (unit=unit1,keyeq=-1,keyid=2,lostat=status) pkey, tch, skey2
0173          else
0174              read (unit=unit1,keyeq=skey1,keyid=1,lostat=status)
0175                  1 pkey, tch, skey2
0176          end if
0177          unlock (unit=unit1)
0178          if ( status .eq. FOR$IOS_SPERECLC ) then                                ! Record lock.
0179              type *, 'Record, status ', skey1, ' locked. Waiting 1 sec.',
0180                  1 uparrow
0181              call lib$wait( 1.0 )
0182              type *, spaces, uparrow
0183              go to 10000
0184          else if ( status .ne. 0 ) then                                        ! Error initial read.
0185              if ( ( skey1 .eq. '*' ) .and. ( sort_open ) ) then
0186                  close (unit=sort_unit)
0187                  call lib$free_lun( sort_unit )
0188                  sort_open = .false.
0189                  stage = -1
0190              end if
0191              next_rec = .false.
0192          else                                                                ! Success first read.
0193              if ( skey1 .eq. '*' ) then
0194                  rd_ptr = 1
0195                  ls_ptr = 1
0196                  write (unit=sort_unit,rec=rd_ptr) pkey
0197              end if
0198              stage = 1
0199              next_rec = .true.

```

NEXT_REC

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```

0200         end if
0201         return
0202     end if
0203
0204     c
0205     if ( stage .eq. 2 ) then
0206         read (unit=sort_unit,rec=rd_ptr) parkey
0207         10100 continue ! "Do while records".
0208         read (unit=unit2,iostat=status) pkey, tch, skey2
0209         unlock (unit=unit2)
0210         if ( status .eq. FOR$IOS_SPERECLOC ) then ! Record lock.
0211             type *, 'Index record locked. Waiting 1 sec.', uparrow
0212             call lib$wait( 1.0 )
0213             type *, spaces, uparrow
0214             go to 10100
0215         else if ( ( status .eq. 0 ) .and. ( skey2 .eq. parkey ) ) then
0216             shf_ptr = ls_ptr
0217             ls_ptr = ls_ptr + 1
0218             do while ( shf_ptr .ge. ins_ptr )
0219                 read (unit=sort_unit,rec=shf_ptr) tpkey, tskey
0220                 write (unit=sort_unit,rec=shf_ptr+1) tpkey, tskey
0221                 shf_ptr = shf_ptr - 1
0222             end do
0223             write (unit=sort_unit,rec=ins_ptr) pkey, skey2
0224             ins_ptr = ins_ptr + 1
0225             go to 10100
0226         end if
0227         rd_ptr = rd_ptr + 1
0228         stage = 1
0229     end if
0230
0231     c
0232     if ( stage .eq. 1 ) then
0233         if ( skey1 .eq. '*' ) read (unit=sort_unit,rec=rd_ptr) tpkey
0234         10200 continue
0235         if ( skey1 .eq. '*' ) then
0236             read (unit=unit2,keyeq=tpkey,keyid=2,iostat=status)
0237             1 pkey, tch, skey2
0238             unlock (unit=unit2)
0239         else
0240             read (unit=unit1,iostat=status) pkey, tch, skey2
0241             unlock (unit=unit1)
0242         end if
0243         if ( status .eq. FOR$IOS_SPERECLOC ) then ! Record lock.
0244             type *, 'Index record locked. Waiting 1 sec.', uparrow
0245             call lib$wait( 1.0 )
0246             type *, spaces, uparrow
0247             go to 10200
0248         else if ( status .eq. 0 ) then
0249             if ( skey1 .eq. '*' ) then
0250                 ins_ptr = rd_ptr + 1
0251                 shf_ptr = ls_ptr
0252                 ls_ptr = ls_ptr + 1
0253                 do while ( shf_ptr .ge. ins_ptr )
0254                     read (unit=sort_unit,rec=shf_ptr) tpkey, tskey
0255                     write (unit=sort_unit,rec=shf_ptr+1) tpkey, tskey
0256                     shf_ptr = shf_ptr - 1
0257                 end do
0258                 write (unit=sort_unit,rec=ins_ptr) pkey

```

NEXT_REC

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[C]

```

0257         ins_ptr = ins_ptr + 1
0258         stage = 2
0259         next_rec = .true.
0260     else
0261         next_rec = tch .eq. skey1
0262     end if
0263 else
0264     if ( skey1 .eq. '*' ) then
0265         if ( rd_ptr .lt. ls_ptr ) then
0266             rd_ptr = rd_ptr + 1
0267             read (unit=sort_unit,recard_ptr) pkey, skey2
0268             next_rec = .true.
0269         else
0270             10300 continue
0271             read (unit=unit1,iostat=status) pkey, tch, skey2
0272             unlock (unit=unit1)
0273             if ( status .eq. FOR$IOS_SPERECLOC ) then ! Record lock.
0274                 type *, 'Index record locked. Waiting 1 sec.', uparrow
0275                 call lib$wait( 1.0 )
0276                 type *, spaces, uparrow
0277                 go to 10300
0278             else if ( ( status .eq. 0 ) .and. ( skey2 .eq. -1 ) ) then
0279                 rd_ptr = 1
0280                 ls_ptr = 1
0281                 write (unit=sort_unit,recard_ptr) pkey
0282                 next_rec = .true.
0283             else
0284                 stage = -1
0285                 next_rec = .false.
0286             end if
0287         end if
0288     else
0289         next_rec = .false.
0290     end if
0291 end if
0292 return
0293 end if
0294 c
0295 end

```

18-Apr-1988 14:00:27 VA
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```

0001 c
0002 c      subroutine submit_job( log_unit, prog_num )
0003 c
0004 c=====
0005 c
0006 c      This subroutine uses function SPAWN_IF to submit, if necessary,
0007 c      the batch job listed in the record whose primary key is prog_num in the
0008 c      index-organization file open on logical unit number log_unit. The
0009 c      parameters log_unit and prog_num are INTEGER*4.
0010 c
0011 c=====
0012 c
0013 c      implicit none
0014 c      external lib$wait, spawn_if
0015 c      include '(system_symbols)'
0183 c      include '6000*RT:[CHUCK.RESEARCH.FORTDIR]SPAWNED_DEF.TXT'
0175 c      include 'SYS$LIBRARY:FORIOSDEF'
0262 c
0263 c      integer*4 spawn_if
0264 c
0265 c      integer*4 log_unit, prog_num
0266 c
0267 c      integer*4 status, ios
0268 c      character*1 uparrow( 3 ) / 27, 'L', 'A' /, spaces( 70 ) / 70* ' ' /
0269 c
0270 c      status = spawn_if( log_unit, prog_num )
0271 c      if ( status .eq. 0 ) return
0272 c
0273 c      10000 continue
0274 c      read (unit=log_unit,keyeq=prog_num,keyid=0,ios=ios)
0275 c      1 sp_num, sp_stat, sp_file
0276 c      unlock (unit=log_unit)
0277 c      if ( ios .eq. FOR$IOS_SPERECLOC ) then ! Record lock.
0278 c      type *, 'Spawn record #', prog_num, ' locked. Waiting 1 sec.',
0279 c      1 uparrow
0280 c      call lib$wait( 1.0 ) ! Wait 1 sec.
0281 c      type *, spaces, uparrow
0282 c      go to 10000
0283 c      else if ( ios .ne. 0 ) then ! Error read.
0284 c      type *
0285 c      type *, 'Error reading spawned-processes index file, code ', ios, '.'
0286 c      type *, 'Couldn't read record with primary key value ', prog_num,
0287 c      1 ''
0288 c      type *
0289 c      return
0290 c      end if
0291 c
0292 c      type *
0293 c      type *, 'Couldn't execute file:'
0294 c      type *, sp_file
0295 c      type *
0296 c      if ( status .eq. 1 ) then
0297 c      type *, 'Error reading spawned-processes index file.'
0298 c      else if ( status .eq. 2 ) then
0299 c      type *, 'Error spawning subprocess.'
0300 c      else if ( status .eq. 3 ) then
0301 c      type *, 'Error rewriting spawned-processes index file.'

```

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```

0302         else
0303             type *, 'Impossible error, number ', status, '.'
0304         end if
0305     c
0306         return
0307     end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	600	PIC CON REL LCL SHR EXE RD NOWRT LONG
1 \$PDATA	297	PIC CON REL LCL SHR NOEXE RD NOWRT LONG
2 \$LOCAL	312	PIC CON REL LCL NOSHR NOEXE RD WRT LONG
Total Space Allocated	1209	

ENTRY POINTS

Address	Type	Name
0-00000000		SUBMIT_JOB

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type	Name
**	1*4	IOS	AP-00000004@	1*4	LOG_UNIT	AP-00000008@	1*4	PROG_NUM
2-0000009C	1*4	SP_NUM	**		CHAR SP_PROC	2-000000A0	1*4	SP_STAT

ARRAYS

Address	Type	Name	Bytes	Dimensions
2-00000003	CHAR	SPACES	70	(70)
2-00000000	CHAR	UPARROW	3	(3)

LABELS

Address	Label
0-00000028	10000

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```

0001 c
0002 c      subroutine print_exp( exp_num )
0003 c
0004 c=====
0005 c
0006 c      This subroutine spawns a subprocess which prints the output
0007 c      file for the MUSIC experiment selected by the INTEGER*4 value exp_num.
0008 c
0009 c=====
0010 c
0011 c      implicit none
0012 c      external lib$find_file, lib$find_file_end, lib$spawn
0013 c
0014 c      integer*4 lib$find_file, lib$find_file_end, lib$spawn
0015 c
0016 c      integer*4 exp_num
0017 c
0018 c      integer*4 status, scratch
0019 c      character*10 expm_sel
0020 c      character*255 scr_str
0021 c
0022 c      scratch = 0
0023 c      write (unit=expm_sel,fmt=90000) exp_num
0024 c      status = lib$find_file( '6000*RT:[CHUCK.PARAMS]MUSIC_OUTPUT'
0025 c      1 // expm_sel // '.DAT', scr_str, scratch )
0026 c      if ( status ) then
0027 c          status = lib$spawn( '$ print 6000*RT:[CHUCK.PARAMS]MUSIC_OUTPUT'
0028 c      1 // expm_sel // '.DAT' )
0029 c          if ( .not. status ) go to 10000
0030 c      else
0031 c      10000 continue
0032 c          type *
0033 c          type *, char( 7 ), 'Can''t print an output file for experiment #',
0034 c      1 expm_sel, '.'
0035 c          type *, '(Output file in use by another program?)'
0036 c          type *
0037 c      end if
0038 c
0039 c      if ( .not. lib$find_file_end( scratch ) ) then
0040 c          type *
0041 c          type *, char( 7 ), 'Error on call to LIB$FIND_FILE_END.'
0042 c          type *
0043 c      end if
0044 c
0045 c      return
0046 c
0047 c      90000 format( i10.10 )
0048 c
0049 c      end

```

18-Apr-1988 14:06:48
2-Apr-1988 14:48:17

```

0001      program sampl_music
0002      c
0003      c ::::::::::::::::::::::::::::::::::::::::::::::::::::::::::::
0004      c
0005      c          This program lurks in the SYS$SLOWBATCH queue (usually), and
0006      c          pounces on any record with a status of ' ' in MUSIC experiment index
0007      c          file 6000$RT:[CHUCK.PARAMS]MUSIC_INDEX.DAT. When it finds such a
0008      c          record, it reads the information necessary to generate sample sets from
0009      c          the appropriate data files, and generates samples for the experiment
0010      c          specified in the selected index file record. The program then writes
0011      c          its results to output files (selected by the experiment number), and
0012      c          updates the MUSIC experiment index record to a status of 'I'. The
0013      c          above sequence is repeated until no more ' ' status records are found
0014      c          AND no process named "Chuck MUSIC Drv" can be found on the VAX.
0015      c          The above description applies exactly to error-free operation
0016      c          only. If a computational error occurs during sample generation, or a
0017      c          file error occurs during writing of the solutions, the program changes
0018      c          the index record's status to 'X' instead of 'I'. If a file error
0019      c          occurs during reading of the input data, the program increments the
0020      c          value mi_ierr (fourth field in the index file record). If the mi_ierr
0021      c          value then exceeds 10, the index record's status is changed to 'X'
0022      c          rather than 'I'. If the mi_ierr value does not exceed 10, the index
0023      c          record's status is changed to '-' rather than 'I' (and the index
0024      c          record's mi_ierr value is updated).
0025      c          NOTE that, since only "ancestor" experiments are ever given the
0026      c          status ' ', only "ancestor" experiments will ever be processed by this
0027      c          program.
0028      c
0029      c ::::::::::::::::::::::::::::::::::::::::::::::::::::::::::::
0030      c
0031      c          implicit none
0032      c          intrinsic dsqrt, dble
0033      c          external lib$get_lun, lib$stop, lib$wait, lib$delete_file
0034      c          external init_va, open_va, exit_va, dump_va
0035      c          external readarr, writearr
0036      c          external music_init, sigmat_form, music_inp_rep, sigvec, obsvec
0037      c          external c16_vec_ins, c16_col_ins, dc_matcopy, c16_mat_dr_leftmul
0038      c          external sqdc_matclr, dc_diad, sqdc_matadd, c16_sel_mdifff
0039      c          external c16_vec_normsq, sqdc_mat_intdiv, dc_mathermmpy
0040      c          external c16_mat_dr_leftdiv, music_out_rep, dc_matconjg, va_dc_qr
0041      c          external endspawn_if, spawn_if
0042      c
0043      c          real*8 c16_vec_normsq
0044      c
0045      c          include '6000$RT:[CHUCK.RESEARCH.FORTDIR]VATYPES.TXT'
0057      c          include '6000$RT:[CHUCK.RESEARCH.FORTDIR]MUSIC_MEASURES.FOR'
0146      c          include '6000$RT:[CHUCK.RESEARCH.FORTDIR]MUSIC_INDEX_DEF.TXT'
0287      c          include '6000$RT:[CHUCK.RESEARCH.FORTDIR]SPAWNED_DEF.TXT'
0380      c          include 'SYS$LIBRARY:FORIOSDEF'
0416      c
0417      c          logical readarr, writearr, endspawn_if
0418      c          integer*4 lib$get_lun, spawn_if
0419      c          integer*4 my_prog, next_prog
0420      c          parameter (my_prog = 1)
0421      c          parameter (next_prog = 2)
0422      c
0423      c          integer*4 status, index_unit, data_unit, spawn_unit

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VAX

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0424      integer*4 waited, wait_lim/ 4 /, vstatus
0425      integer*4 ants_by_2( 2 ), waves_by_9( 2 ), waves_by_2( 2 )
0426      integer*4 nois_by_2( 2 ), nois_by_3( 2 ), ants_by_nois( 2 )
0427      integer*4 waves_lin( 1 ), ants_by_wvs( 2 ), ants_lin( 1 )
0428      integer*4 smp_by_ants( 2 ), wavs_by_smp( 2 ), ants_by_smp( 2 )
0429      integer*4 anois_by_2( 2 ), anois_by_3( 2 ), ants_by_anois( 2 )
0430      integer*4 ants_sqr( 2 ), robs_add, rb_add, ants_di_add, rnoise_add
0431      integer*4 arnoise_add, anoise_add, obst_add, noiset_add
0432      integer*4 ant_add, wave_add, mtype_add, ndef_add, npars_add, correl_add
0433      integer*4 ccamp_add, sgmt_add, sspc_add, cln_add, obs_add, noise_add
0434      integer*4 obsmat_add, noisemat_add, sigvm_add, clnmat_add
0435      integer*4 andef_add, anpars_add, acorrel_add
0436      integer*4 i
0437      real*8 sum_normsq, lambda_min
0438      character*25 work_file
0439      character*80 msg
0440      c
0441      c          Get logical unit numbers
0442      c
0443      status = lib$get_lun( index_unit )
0444      if (.not. status) call lib$stop( Xval( status ) )
0445      status = lib$get_lun( data_unit )
0446      if (.not. status) call lib$stop( Xval( status ) )
0447      status = lib$get_lun( spawn_unit )
0448      if (.not. status) call lib$stop( Xval( status ) )
0449      c
0450      c          Open the MUSIC index file.
0451      c
0452      10000      continue
0453      open (unit=index_unit,file='6000*RT:[CHUCK.PARAMS]MUSIC_INDEX.DAT',
0454           1      form='UNFORMATTED',recordtype='FIXED',recl=64,
0455           2      organization='INDEXED',access='KEYED',status='OLD',
0456           3      key=(1:4:INTEGER,5:5:CHARACTER),dispose='KEEP',SHARED,
0457           4      err=10100)
0458      go to 10110
0459      c
0460      c          Error opening the MUSIC index file or MUSIC spawned-processes
0461      c          index file; abandon hope.
0462      c
0463      10100      continue
0464      type *
0465      type *, 'Error opening MUSIC index file or spawned-processes file.'
0466      go to 60620
0467      c
0468      c          MUSIC index file successfully opened.  Open MUSIC spawned-
0469      c          processes index file.
0470      c
0471      10110      continue
0472      open (unit=spawn_unit,file='6000*RT:[CHUCK.PARAMS]SPAWNED.IDX',
0473           1      form='UNFORMATTED',recordtype='FIXED',recl=26,
0474           2      organization='INDEXED',access='KEYED',status='OLD',
0475           3      key=(1:4:INTEGER,5:8:INTEGER),dispose='KEEP',SHARED,err=10100)
0476      c
0477      c          MUSIC index file and MUSIC spawned-processes index file
0478      c          successfully opened.  Start lurking, waiting to pounce on unsuspecting
0479      c          status ' ' records.
0480      c

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0481      waited = 0
0482      10120      continue
0483      read (unit=index_unit,keyeq=' ',keyid=1,iostat=status,err=20100)
0484      1          mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
0485      2          mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
0486      3          mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
0487      4          mi_tol, mi_which, mi_ofile, mi_anoises, mi_irlslm,
0488      5          mi_zeigs, mi_doas, mi_estype, mi_parp
0489      unlock (unit=index_unit)
0490      waited = 0
0491      go to 20110
0492      c
0493      c          Error reading status ' ' records.  Could be that the first such
0494      c          record is locked, no such records exist, or some other error.
0495      c
0496      20100      continue
0497      unlock (unit=index_unit)
0498      if ( status .eq. FOR$IOS_SPERECLC ) then          ! Locked.
0499      type *, 'Record locked, status ' ' '.  Waiting 5 min.'
0500      waited = 0
0501      20120      continue          ! Branch to wait then read.
0502      call lib$wait( 300.0 )          ! Hibernate for 5 minutes.
0503      go to 10120
0504      else if ( status .eq. FOR$IOS_ATTACCNON ) then          ! No record.
0505      go to 70000
0506      else          ! Some other error.
0507      type *
0508      type *, 'Failed reading status ' ' ' Index records.'
0509      call lib$stop( %val( status ) )
0510      and if
0511      c
0512      c          Successfully read a status ' ' record into the batch of
0513      c          variables listed in the read statement above.  Commence sample
0514      c          generation.
0515      c          Start by beating the Virtual Array system (copyright 1987 by
0516      c          Dwight Day) to death.
0517      c
0518      20110      continue
0519      c
0520      call init_va
0521      ants_by_2(1) = mi_ants
0522      ants_by_2(2) = 2
0523      waves_by_9(1) = mi_waves
0524      waves_by_9(2) = 9
0525      waves_by_2(1) = mi_waves
0526      waves_by_2(2) = 2
0527      waves_l1n(1) = mi_waves
0528      ants_by_wvs(1) = mi_ants
0529      ants_by_wvs(2) = mi_waves
0530      ants_l1n(1) = mi_ants
0531      ants_sqr(1) = mi_ants
0532      ants_sqr(2) = mi_ants
0533      smp_by_ants(1) = mi_smp
0534      smp_by_ants(2) = mi_ants
0535      wavs_by_smp(1) = mi_waves
0536      wavs_by_smp(2) = mi_smp
0537      ants_by_smp(1) = mi_ants

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0538      ants_by_smp(2) = mi_smp
0539      nois_by_2(1)   = mi_noises
0540      nois_by_2(2)   = 2
0541      nois_by_3(1)   = mi_noises
0542      nois_by_3(2)   = 3
0543      ants_by_nois(1) = mi_ants
0544      ants_by_nois(2) = mi_noises
0545      if ( mi_which .gt. 0 ) then
0546          anois_by_2(1) = mi_anoises
0547          anois_by_2(2) = 2
0548          anois_by_3(1) = mi_anoises
0549          anois_by_3(2) = 3
0550          ants_by_anois(1) = mi_ants
0551          ants_by_anois(2) = mi_anoises
0552      end if
0553
0554      call open_va( ants_di_add, ants_sqr, 2, dc_type, vstatus, msg )
0555      call exit_va( vstatus, msg )
0556      call open_va( ant_add, ants_by_2, 2, dr_type, vstatus, msg )
0557      call exit_va( vstatus, msg )
0558      call open_va( wave_add, waves_by_9, 2, dr_type, vstatus, msg )
0559      call exit_va( vstatus, msg )
0560      call open_va( mtype_add, waves_by_2, 2, di_type, vstatus, msg )
0561      call exit_va( vstatus, msg )
0562      call open_va( ndef_add, nois_by_2, 2, di_type, vstatus, msg )
0563      call exit_va( vstatus, msg )
0564      call open_va( npars_add, nois_by_3, 2, sr_type, vstatus, msg )
0565      call exit_va( vstatus, msg )
0566      call open_va( correl_add, ants_by_nois, 2, dr_type, vstatus, msg )
0567      call exit_va( vstatus, msg )
0568      if ( mi_which .gt. 0 ) then
0569          call open_va( andef_add, anois_by_2, 2, di_type, vstatus, msg )
0570          call exit_va( vstatus, msg )
0571          call open_va( anpars_add, anois_by_3, 2, sr_type, vstatus, msg )
0572          call exit_va( vstatus, msg )
0573          call open_va( acorrel_add, ants_by_anois, 2,
0574                      dr_type, vstatus, msg )
0575          call exit_va( vstatus, msg )
0576          call open_va( anoise_add, ants_lin, 1, dc_type, vstatus, msg )
0577          call exit_va( vstatus, msg )
0578          call open_va( arnoise_add, ants_sqr, 2, dc_type, vstatus, msg )
0579          call exit_va( vstatus, msg )
0580      end if
0581      call open_va( ccamp_add, waves_lin, 1, dc_type, vstatus, msg )
0582      call exit_va( vstatus, msg )
0583      call open_va( sgmt_add, ants_by_wvs, 2, dc_type, vstatus, msg )
0584      call exit_va( vstatus, msg )
0585      call open_va( sspc_add, waves_lin, 1, dc_type, vstatus, msg )
0586      call exit_va( vstatus, msg )
0587      call open_va( cln_add, ants_lin, 1, dc_type, vstatus, msg )
0588      call exit_va( vstatus, msg )
0589      call open_va( obs_add, ants_lin, 1, dc_type, vstatus, msg )
0590      call exit_va( vstatus, msg )
0591      call open_va( robs_add, ants_sqr, 2, dc_type, vstatus, msg )
0592      call exit_va( vstatus, msg )
0593      call open_va( obst_add, ants_sqr, 2, dc_type, vstatus, msg )
0594      call exit_va( vstatus, msg )

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0595      call open_va( noise_add, ants_lin, 1, dc_type, vstatus, msg )
0596      call exit_va( vstatus, msg )
0597      call open_va( rnoise_add, ants_sqr, 2, dc_type, vstatus, msg )
0598      call exit_va( vstatus, msg )
0599      call open_va( rb_add, ants_sqr, 2, dc_type, vstatus, msg )
0600      call exit_va( vstatus, msg )
0601      call open_va( noiset_add, ants_sqr, 2, dc_type, vstatus, msg )
0602      call exit_va( vstatus, msg )
0603      call open_va( obsmat_add, smp_by_ants, 2, dc_type, vstatus, msg )
0604      call exit_va( vstatus, msg )
0605      call open_va( noisemat_add, smp_by_ants, 2, dc_type, vstatus, msg )
0606      call exit_va( vstatus, msg )
0607      call open_va( sigvm_add, wavs_by_smp, 2, dc_type, vstatus, msg )
0608      call exit_va( vstatus, msg )
0609      call open_va( climat_add, ants_by_smp, 2, dc_type, vstatus, msg )
0610      call exit_va( vstatus, msg )
0611      c
0612      c          Get problem definition from input file, initialize the noise
0613      c          counters, and form the array phase shift matrix.
0614      c
0615      work_file = 'MUSIC_INPUT_____DAT'
0616      write (unit=work_file( 12 : 21 ),fmt=90020,err=60510) mi_pkey
0617      c
0618      open (unit=data_unit,file='6000%RT:[CHUCK.PARAMS]//work_file,
0619      1      status='OLD',organization='SEQUENTIAL',access='SEQUENTIAL',
0620      2      form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
0621      3      READONLY,err=60510)
0622      read (unit=data_unit,err=60500)
0623      if ( .not. (
0624      1      readarr( %val( ant_add ), dr_type, mi_ants, 2, data_unit, 32 )
0625      2      ) ) go to 60500
0626      if ( .not. (
0627      1      readarr( %val( wave_add ), dr_type, mi_waves, 9, data_unit,
0628      2      32 )
0629      3      ) ) go to 60500
0630      if ( .not. (
0631      1      readarr( %val( mtype_add ), di_type, mi_waves, 2, data_unit,
0632      2      64 )
0633      3      ) ) go to 60500
0634      if ( .not. (
0635      1      readarr( %val( ndef_add ), di_type, mi_noises, 2, data_unit,
0636      2      64 )
0637      3      ) ) go to 60500
0638      if ( .not. (
0639      1      readarr( %val( npars_add ), sr_type, mi_noises, 3, data_unit,
0640      2      64 )
0641      3      ) ) go to 60500
0642      if ( .not. (
0643      1      readarr( %val( correl_add ), dr_type, mi_ants, mi_noises,
0644      2      data_unit, 32 )
0645      3      ) ) go to 60500
0646      if ( mi_which .gt. 0 ) then
0647      if ( .not. (
0648      1      readarr( %val( andef_add ), di_type, mi_anosies, 2,
0649      2      data_unit, 64 )
0650      3      ) ) go to 60500
0651      if ( .not. (

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0652      1      readarr( %val( anpars_add ), sr_type, mi_noises, 3,
0653      2      data_unit, 64 )
0654      3      ) ) go to 6050C
0655      if ( .not. (
0656      1      readarr( %val( acorrel_add ), dr_type, mi_ants, mi_noises,
0657      2      data_unit, 32 )
0658      3      ) ) go to 6050C
0659      end if
0660      close (unit=data_unit)
0661      c
0662      call music_init( mi_waves, %val( wave_add ), %val( ccamp_add ),
0663      1      mi_noises, %val( ndef_add ), %val( npars_add ),
0664      2      mi_noises, %val( andef_add ), %val( anpars_add ),
0665      3      mi_which )
0666      call sigmat_form( %val( ant_add ), mi_ants,
0667      1      %val( wave_add ), mi_waves, %val( sgmt_add ) )
0668      c
0669      c      Print out input parameters (converted as appropriate).
0670      c
0671      call music_inp_rep( data_unit, mi_pkey, mi_ofile, mi_dtim,
0672      1      %val( ant_add ), mi_ants, %val( wave_add ), %val( mtype_add ),
0673      2      mi_waves, %val( ndef_add ), %val( npars_add ),
0674      3      %val( correl_add ), mi_noises, mi_which, %val( andef_add ),
0675      4      %val( anpars_add ), %val( acorrel_add ), mi_noises, mi_smp,
0676      5      mi_sint, mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_tol,
0677      6      mi_irlslim )
0678      c
0679      c      Collect noise and observation sample vectors into their
0680      c      respective matrices, and accumulate the actual noise covariance
0681      c      matrix if necessary.
0682      c
0683      sum_normsq = 0.0d0
0684      if ( mi_which .gt. 0 )
0685      1      call sqdc_matclr( %val( arnoise_add ), mi_ants )
0686      c
0687      do i = 1, mi_smp
0688      call sigvec( mi_waves, %val( ccamp_add ), %val( wave_add ),
0689      1      %val( mtype_add ), i, mi_sint, %val( sspc_add ) )
0690      call obsvec( %val( sgmt_add ), mi_ants, mi_waves,
0691      1      %val( ndef_add ), %val( npars_add ),
0692      2      %val( correl_add ), mi_noises, %val( andef_add ),
0693      3      %val( anpars_add ), %val( acorrel_add ), mi_noises,
0694      4      %val( sspc_add ), %val( cln_add ), %val( obs_add ),
0695      5      %val( noise_add ), mi_which )
0696      call c16_vec_ins( %val( obs_add ), mi_ants, %val( obsmat_add ),
0697      1      mi_smp, mi_ants, i )
0698      call c16_vec_ins( %val( noise_add ), mi_ants,
0699      1      %val( noisemat_add ), mi_smp, mi_ants, i )
0700      call c16_col_ins( %val( sspc_add ), mi_waves, %val( sigvm_add ),
0701      1      mi_waves, mi_smp, i )
0702      call c16_col_ins( %val( cln_add ), mi_ants, %val( clnmat_add ),
0703      1      mi_ants, mi_smp, i )
0704      c
0705      if ( mi_which .gt. 0 ) then
0706      call c16_sel_mdif( %val( obs_add ), 1, 1, %val( cln_add ),
0707      1      1, 1, mi_ants, %val( anoise_add ) )
0708      call dc_diad( %val( anoise_add ), mi_ants,

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0709          1                                %val( ants_di_add ) )
0710          call sqdc_matadd( %val( arnoise_add ), %val( ants_di_add ),
0711          1                                mi_ants )
0712          end if
0713          c
0714          sum_normsq = sum_normsq
0715          1                                + c16_vec_normsq( %val( noise_add ), mi_ants
0716          end do
0717          c
0718          c                                Write the generated vector samples to the "samples" file for
0719          c                                this experiment.
0720          c
0721          work_file( 7 : 11 ) = 'SAMPL'
0722          call lib$delete_file( '6000$RT:[CHUCK.PARAMS]' // work_file // ',*'
0723          open (unit=data_unit,file='6000$RT:[CHUCK.PARAMS]//work_file,
0724          1                                status='NEW',organization='SEQUENTIAL',access='SEQUENTIAL',
0725          2                                form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
0726          3                                err=60530)
0727          if ( .not. (
0728          1                                writearr( %val( obsmat_add ), dc_type, mi_smp, mi_ants,
0729          2                                data_unit, 16
0730          3                                ) ) go to 605
0731          if ( .not. (
0732          1                                writearr( %val( noisemat_add ), dc_type, mi_smp, mi_ants,
0733          2                                data_unit, 16
0734          3                                ) ) go to 605
0735          if ( .not. (
0736          1                                writearr( %val( sigvm_add ), dc_type, mi_waves, mi_smp,
0737          2                                data_unit, 16
0738          3                                ) ) go to 605
0739          if ( .not. (
0740          1                                writearr( %val( cinmat_add ), dc_type, mi_ants, mi_smp,
0741          2                                data_unit, 16
0742          3                                ) ) go to 605
0743          close (unit=data_unit)
0744          c
0745          c                                Compute minimum generalized eigenvalue estimate and covarianc
0746          c                                matrices.
0747          c
0748          lambda_min = sum_normsq / ( mi_ants * mi_smp )
0749          if ( mi_which .gt. 0 )
0750          1                                call sqdc_mat_intdiv( %val( arnoise_add ), mi_ants, mi_smp )
0751          c
0752          c                                Form normalized complex conjugates of observation and noise
0753          c                                sample matrices, then perform QR decomposition of normalized complex
0754          c                                conjugates.
0755          c
0756          call dc_matconjg( %val( obsmat_add ), mi_smp, mi_ants )
0757          call c16_mat_dr_leftdiv( %val( obsmat_add ), mi_smp, mi_ants,
0758          1                                dsqrt( dble( mi_smp ) ), mi_ants )
0759          call va_dc_qr( %val( obsmat_add ), mi_smp, mi_ants,
0760          1                                %val( obst_add ) )
0761          c
0762          call dc_matconjg( %val( noisemat_add ), mi_smp, mi_ants )
0763          call c16_mat_dr_leftdiv( %val( noisemat_add ), mi_smp, mi_ants,
0764          1                                dsqrt( sum_normsq / mi_ants ), mi_ants )
0765          call va_dc_qr( %val( noisemat_add ), mi_smp, mi_ants,

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0766                                         1          %val( noiset_add ) :
0767     c
0768     c          Compute observation, normalized (model) noise, and (model)
0769     c noise covariance matrices.
0770     c
0771     c          call dc_mathermmpy( %val( obst_add ), mi_ants, mi_ants,
0772     1          %val( rb_add ), mi_ants, %val( robs_add )
0773     c          call dc_mathermmpy( %val( noiset_add ), mi_ants, mi_ants,
0774     1          %val( noiset_add ), mi_ants, %val( rb_add )
0775     c          call dc_matcopy( %val( rb_add ), mi_ants, mi_ants,
0776     1          %val( rnoise_add )
0777     c          call c16_mat_dr_leftmul( %val( rnoise_add ), mi_ants, mi_ants,
0778     1          lambda_min, mi_ants
0779     c
0780     c          Print out accumulated observation and noise covariance
0781     c matrices.
0782     c
0783     c          call music_out_rep( data_unit, mi_pkey, mi_park, mi_parp, -2,
0784     1          mi_which, mi_ofile, %val( robs_add ),
0785     2          %val( rnoise_add ), %val( arnoise_add ),
0786     3          mi_ants, lambda_min, mi_dtr
0787     c
0788     c          Create the experiment's "eigenvectors" file, and write
0789     c covariance matrices and triangular matrices from QR decompositions
0790     c it.
0791     c
0792     c          work_file( 7 : 11 ) = 'EIGVS'
0793     c          call lib$delete_file( '6000$RT:[CHUCK.PARAMS]' // work_file // ',' )
0794     c          open (unit=data_unit, file='6000$RT:[CHUCK.PARAMS]'//work_file,
0795     1          status='NEW', organization='SEQUENTIAL', access='SEQUENTIAL',
0796     2          form='UNFORMATTED', recordtype='FIXED', recl=64, dispose='KEEP'
0797     3          err=60530)
0798     c          if ( .not. ( writearr( %val( robs_add ), dc_type, mi_ants, mi_ants,
0799     1          data_unit, 16 ) ) ) go to 60
0800     c          if ( .not. ( writearr( %val( rnoise_add ), dc_type, mi_ants,
0801     1          mi_ants, data_unit, 16 ) ) ) go to 60520
0802     c          if ( mi_which .gt. 0 ) then
0803     c          if ( .not. ( writearr( %val( arnoise_add ), dc_type, mi_ants,
0804     1          mi_ants, data_unit, 16 ) ) ) go to 60520
0805     c          end if
0806     c          if ( .not. ( writearr( %val( rb_add ), dc_type, mi_ants, mi_ants,
0807     1          data_unit, 16 ) ) ) go to 60
0808     c          if ( .not. ( writearr( %val( obst_add ), dc_type, mi_ants, mi_ants,
0809     1          data_unit, 16 ) ) ) go to 60
0810     c          if ( .not. ( writearr( %val( noiset_add ), dc_type, mi_ants,
0811     1          mi_ants, data_unit, 16 ) ) ) go to 60520
0812     c          close (unit=data_unit)
0813     c
0814     c          Update index file entry to reflect creation of the "samples"
0815     c and "eigenvectors" files, discard memory used for Virtual Arrays, ar
0816     c exit if there is no more work to do.
0817     c
0818     c          call dump_va( vstatus, msg )
0819     c          call exit_va( vstatus, msg )
0820     c          mi_stat = 'I'
0821     c
0822     c          status = spawn_if( spawn_unit, next_prog ) ! Start eigenvalues

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0823      if ( status .ne. 0 ) then                ! Spawn failed.
0824      type *
0825      type *, 'Error initiating eigenvalue solver.'
0826      if ( status .eq. 1 ) then
0827      type *, 'Read error on spawned-processes index file.'
0828      else if ( status .eq. 2 ) then
0829      type *, 'Error spawning subprocess.'
0830      else if ( status .eq. 3 ) then
0831      type *, 'Error rewriting spawned-processes index file.'
0832      else
0833      type *, 'Impossible error, code ', status, ' .'
0834      end if
0835      60620  continue                             ! Error exit.
0836      type *, 'Sample generation abandoned.'
0837      type *
0838      stop
0839      end if
0840      c
0841      60600  continue
0842      read (unit=index_unit,key=mi_pkey,keyld=0,iostat=status)
0843      if ( status .eq. FOR$IOS_SPERECLOC ) then
0844      type *, 'Record to rewrite locked, #', mi_pkey,
0845      1      ' . Waiting 10 sec.'
0846      unlock (unit=index_unit)
0847      call lib$wait( 10.0 )                       ! Wait 10 seconds.
0848      go to 60600
0849      else if ( status .ne. 0 ) then              ! Error reading index.
0850      60610  continue                             ! Index access error branch.
0851      unlock (unit=index_unit)
0852      type *
0853      type *, 'Error accessing MUSIC experiment index, code ', status
0854      go to 60620
0855      end if
0856      rewrite (unit=index_unit,iostat=status,err=60610)
0857      1      mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
0858      2      mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
0859      3      mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
0860      4      mi_tol, mi_which, mi_ofile, mi_anoises, mi_irlslm,
0861      5      mi_zeigs, mi_doas, mi_estype, mi_parp
0862      unlock (unit=index_unit)
0863      go to 10120
0864      c
0865      c      No more status ' ' records available.
0866      c
0867      70000  continue
0868      if ( waited .ge. wait_lim ) then          ! Already waited 20 minutes.
0869      waited = 0
0870      if ( endspawn_if( spawn_unit, my_prog ) ) stop
0871      else
0872      waited = waited + 1
0873      call lib$wait( 300.0 )                     ! Wait 5 minutes & try again.
0874      end if
0875      go to 10120
0876      c
0877      c*****
0878      c
0879      c      File error processing branches.

```

SAMPL_MUSIC

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```
0880 c
0881 c*****
0882 c
0883 c          Computation error, or error opening or writing samples output
0884 c          or eigenvectors output file.
0885 c
0886 60520 continue
0887          close (unit=data_unit)
0888 c
0889 60530 continue
0890          call dump_va( vstatus, msg )
0891          call exit_va( vstatus, msg )
0892 c
0893 60540 continue
0894          mi_stat = 'X'
0895          go to 60600
0896 c
0897 c          Error reading input data file.
0898 c
0899 60500 continue
0900          close (unit=data_unit)
0901 c
0902 60510 continue
0903          call dump_va( vstatus, msg )
0904          call exit_va( vstatus, msg )
0905 c
0906          if ( mi_ierr .ge. 10 ) then
0907              mi_stat = 'X'
0908          else
0909              mi_ierr = mi_ierr + 1
0910              mi_stat = '-'
0911          end if
0912          go to 60600
0913 c
0914 90020 format( i10.10 )
0915 c
0916          end
```


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```

0001      program eig_music
0002      c
0003      c:.....
0004      c
0005      c      This program lurks in the SYS$SLOWBATCH queue (usually), and
0006      c      pounces on any record with a status of 'I' in MUSIC experiment index
0007      c      file 6000$RT:[CHUCK.PARAMS]MUSIC_INDEX.DAT. When it finds such a
0008      c      record, it reads the information necessary to solve the generalized
0009      c      eigenvalue problem from the appropriate data files, and proceeds to
0010      c      solve the problem for the experiment specified in the selected index
0011      c      file record. The program then writes its results to an eigenvalue dat
0012      c      output file (selected by the experiment number), and updates the MUSIC(
0013      c      experiment index record to a status of 'E'. The above sequence is
0014      c      repeated until no more status 'I' records are found AND no process
0015      c      named "Chuck MUSIC Drv" can be found on the VAX.
0016      c
0017      c      The above description applies exactly to error-free operation
0018      c      only. If a computational error occurs during eigenvalue problem
0019      c      solution, or a file error occurs during writing of the solutions, the
0020      c      program changes the index record's status to 'X' instead of 'E'. If a
0021      c      file error occurs during reading of the input data, the program
0022      c      increments the value mi_eerr (fifth field in the index file record).
0023      c      If the mi_eerr value then exceeds 10, the index record's status is
0024      c      changed to 'X' rather than 'E'. If the mi_eerr value does not exceed
0025      c      10, the index record's status is changed to ' ' (for "ancestor"
0026      c      experiments) or 'D' (for "descendant" experiments) rather than 'E' (ar
0027      c      the index record's mi_eerr value is updated). Finally, the program
0028      c      changes the index record's status to 'I' if a data file can't be opene
0029      c      (the assumption being that the file is in use by another program).
0030      c      Status 'I' records are processed after all status 'I' record operator
0031      c      have been completed.
0032      c:.....
0033      c
0034      c      implicit none
0035      c      external lib$get_lun, lib$stop, lib$wait
0036      c      external open_va, exit_va, dump_va
0037      c      external readarr, writearr, skiparr
0038      c      external sq_vlgsvd, sort_eig, music_eig_rep, vl_check
0039      c      external endspawn_if
0040      c
0041      c      include '6000$RT:[CHUCK.RESEARCH.FORTDIR]IVATYPES.TXT'
0053      c      include '6000$RT:[CHUCK.RESEARCH.FORTDIR]MUSIC_MEASURES.FOR'
0142      c      include '6000$RT:[CHUCK.RESEARCH.FORTDIR]MUSIC_INDEX_DEF.TXT'
0283      c      include '6000$RT:[CHUCK.RESEARCH.FORTDIR]SPAWNED_DEF.TXT'
0326      c      include 'SYS$LIBRARY:FORIOSDEF'
0412      c
0413      c      integer*4 my_prog
0414      c      parameter (my_prog = 2)
0415      c
0416      c      logical readarr, writearr, skiparr, endspawn_if
0417      c      integer*4 lib$get_lun
0418      c
0419      c      logical ancestor
0420      c      integer*4 status, index_unit, data_unit, spawn_unit
0421      c      integer*4 waited, wait_lim/ 4 /, vstatus
0422      c      integer*4 ants_sqr( 2 ), ants_lin( 1 )
0423      c      integer*4 robs_add, rb_add, obst_add, noiset_add, eig_add, eigv_add

```

EIG_MUSIC

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```

0424         integer*4 panic
0425         real*8  condit
0426         character*1 sch_stat
0427         character*25 work_file
0428         character*80 msg
0429         c
0430         c           Get logical unit numbers
0431         c
0432         status = lib$get_lun( index_unit )
0433         if (.not. status) call lib$stop( %val( status ) )
0434         status = lib$get_lun( data_unit )
0435         if (.not. status) call lib$stop( %val( status ) )
0436         status = lib$get_lun( spawn_unit )
0437         if (.not. status) call lib$stop( %val( status ) )
0438         c
0439         c           Open the MUSIC index file.
0440         c
0441         10000  continue
0442         open (unit=index_unit,file='6000*RT:[CHUCK.PARAMS]MUSIC_INDEX.DAT',
0443             1      form='UNFORMATTED',recordtype='FIXED',recl=64,
0444             2      organization='INDEXED',access='KEYED',status='OLD',
0445             3      key=(1:4:INTEGER,5:5:CHARACTER,6:9:INTEGER),
0446             4      dispose='KEEP',SHARED,err=10100)
0447         go to 10110
0448         c
0449         c           Error opening the MUSIC index file or MUSIC spawned-processes
0450         c           index file; abandon hope.
0451         c
0452         10100  continue
0453         type *
0454         type *, 'Error opening MUSIC index file or spawned-processes file.'
0455         go to 60620
0456         c
0457         c           MUSIC index file successfully opened. Open MUSIC spawned-
0458         c           processes index file.
0459         c
0460         10110  continue
0461         open (unit=spawn_unit,file='6000*RT:[CHUCK.PARAMS]SPAWNED.IDX',
0462             1      form='UNFORMATTED',recordtype='FIXED',recl=26,
0463             2      organization='INDEXED',access='KEYED',status='OLD',
0464             3      key=(1:4:INTEGER,5:8:INTEGER),dispose='KEEP',SHARED,err=10100)
0465         c
0466         c           MUSIC index file and MUSIC spawned-processes index file
0467         c           successfully opened. Start lurking, waiting to pounce on unsuspecting
0468         c           status '1' records.
0469         c
0470         waited = 0
0471         sch_stat = '1'
0472         10120  continue
0473         read (unit=index_unit,keyeq=sch_stat,keyid=1,iostat=status,err=20100)
0474         1      mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
0475         2      mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
0476         3      mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
0477         4      mi_tol, mi_which, mi_ofile, mi_anaises, mi_irlslim,
0478         5      mi_zeigs, mi_doas, mi_estype, mi_parp
0479         unlock (unit=index_unit)
0480         sch_stat = '1'

```

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```

0481          waited = 0
0482          go to 21110
0483      c
0484      c          Error reading status 'I' records. Could be that the first such
0485      c          record is locked, no such records exist, or some other error.
0486      c
0487      20100      continue
0488      c          unlock (unit=index_unit)
0489      c          if ( status .eq. FOR$IOS_SPERECLOC ) then          ! Locked.
0490      c              waited = 0
0491      c              type *, 'Locked record, status ', sch_stat, '. Waiting 5 min.'
0492      c              20120      continue          ! Branch to wait then read.
0493      c              call lib$wait( 300.0 )          ! Hibernate for 5 minutes.
0494      c              go to 10120
0495      c          else if ( status .eq. FOR$IOS_ATTACCNON ) then          ! No record.
0496      c              if ( sch_stat .eq. 'I' ) then          ! May be 'I's.
0497      c                  sch_stat = 'I'
0498      c                  go to 10120
0499      c              else          ! Neither 'I's nor 'I's.
0500      c                  go to 70000
0501      c              end if
0502      c          else          ! Some other error.
0503      c              type *
0504      c              type *, 'Failed reading status ''I'' index records.'
0505      c              call lib$stop( %val( status ) )
0506      c          end if
0507      c
0508      c          Successfully read a status 'I' record into the batch of
0509      c          variables listed in the read statement above. Commence eigenvalue
0510      c          problem solution.
0511      c          Start by beating the Virtual Array system (copyright 1987 by
0512      c          Dwight Day) to death.
0513      c
0514      21110      continue
0515      c          ancestor = ( mi_park .eq. -1 )
0516      c
0517      c          call init_va
0518      c          ants_lin(1) = mi_ants
0519      c          ants_sqr(1) = mi_ants
0520      c          ants_sqr(2) = mi_ants
0521      c
0522      c          call open_va( robs_add, ants_sqr, 2, dc_type, vstatus, msg )
0523      c          call exit_va( vstatus, msg )
0524      c          call open_va( rb_add, ants_sqr, 2, dc_type, vstatus, msg )
0525      c          call exit_va( vstatus, msg )
0526      c          call open_va( obst_add, ants_sqr, 2, dc_type, vstatus, msg )
0527      c          call exit_va( vstatus, msg )
0528      c          call open_va( noiset_add, ants_sqr, 2, dc_type, vstatus, msg )
0529      c          call exit_va( vstatus, msg )
0530      c          call open_va( eig_add, ants_lin, 1, dr_type, vstatus, msg )
0531      c          call exit_va( vstatus, msg )
0532      c          call open_va( eigv_add, ants_sqr, 2, dc_type, vstatus, msg )
0533      c          call exit_va( vstatus, msg )
0534      c
0535      c          Get covariance matrices and their QR decompositions from the
0536      c          experiment's "eigenvectors" file, LEAVING THE FILE OPEN for subsequent
0537      c          write of eigenproblem solutions.

```

EIG_MUSIC

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```

0538 c
0539 work_file = 'MUSIC_EIGVS_____DAT'
0540 write (unit=work_file( 12 : 21 ),fmt=90020,err=62510) mi_pkey
0541 c
0542 open (unit=data_unit,file='6000*RT:[CHUCK.PARAMS]//work_file,
0543 1 status='OLD',organization='SEQUENTIAL',access='SEQUENTIAL',
0544 2 form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
0545 3 lostat=status,err=61510)
0546 if ( .not. ( ancestor ) ) then ! Descendant.
0547 if ( .not. ( skiparr( mi_ants, mi_ants,
0548 data_unit, 16 ) ) ) go to 6050
0549 end if
0550 if ( .not. ( readarr( %val( robs_add ), dc_type, mi_ants, mi_ants,
0551 data_unit, 16 ) ) ) go to 6050
0552 if ( ancestor ) then ! Ancestor.
0553 if ( .not. ( skiparr( mi_ants, mi_ants,
0554 data_unit, 16 ) ) ) go to 6050
0555 if ( mi_which .gt. 0 ) then
0556 if ( .not. ( skiparr( mi_ants, mi_ants,
0557 data_unit, 16 ) ) ) go to 6050
0558 end if
0559 end if
0560 if ( .not. ( readarr( %val( rb_add ), dc_type, mi_ants, mi_ants,
0561 data_unit, 16 ) ) ) go to 6050
0562 if ( .not. ( readarr( %val( obst_add ), dc_type, mi_ants, mi_ants,
0563 data_unit, 16 ) ) ) go to 6050
0564 if ( .not. ( readarr( %val( noiset_add ), dc_type, mi_ants,
0565 mi_ants, data_unit, 16 ) ) ) go to 60500
0566 c
0567 c Solve the generalized eigenvalue / eigenvector problem for the
0568 c observation covariance and normalized noise covariance. Solution of
0569 c the eigenvalue / eigenvector problem is via generalized singular value
0570 c decomposition of the normalized observation and noise vector matrices.
0571 c
0572 call sq_vlgsvd( %val( obst_add ), %val( noiset_add ), mi_ants,
0573 1 mi_tol, %val( eig_add ), %val( eigv_add ),
0574 2 condit, panic
0575 call sort_eig( %val( eig_add ), %val( eigv_add ), mi_ants )
0576 c
0577 c Write results to eigenvalue problem output file, and close the
0578 c file.
0579 c
0580 if ( .not. ( writearr( %val( eig_add ), dr_type, mi_ants, 1,
0581 data_unit, 32 ) ) ) go to 6052
0582 if ( .not. ( writearr( %val( eigv_add ), dc_type, mi_ants, mi_ants,
0583 data_unit, 16 ) ) ) go to 6052
0584 close (unit=data_unit)
0585 c
0586 c Print results, check for errors in Van Loan algorithm, and
0587 c discard space used by Virtual Arrays.
0588 c
0589 call music_eig_rep( data_unit, mi_pkey, mi_park, mi_parp,
0590 1 mi_ofile, %val( eig_add ), %val( eigv_add ),
0591 2 mi_ants, mi_dtim )
0592 call vl_check( data_unit, mi_pkey, mi_park, mi_parp, mi_ofile,
0593 1 %val( robs_add ), %val( rb_add ), mi_ants,
0594 2 %val( eig_add ), %val( eigv_add ),

```

```

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0595          3          condit, panic, mi_dtim )
0596          call dump_va( vstatus, msg )
0597          call exit_va( vstatus, msg )
0598          c
0599          if ( panic .ne. 0 ) go to 60530          ! Mark for delete if fail
0600          c
0601          c          Update the experiment's index record, and exit if there is
0602          c more work to do.
0603          c
0604          mi_stat = 'E'
0605          60600          continue
0606          read (unit=index_unit,key=mi_pkey,keyid=0,iostat=status)
0607          if ( status .eq. FOR$IOS_SPERECLC ) then
0608              type *, 'Locked record to rewrite, #', mi_pkey,
0609              1          ' Waiting 10 sec
0610              unlock (unit=index_unit)
0611                  call lib$wait( 10.0 )          ! Wait 10 seconds
0612                  go to 60600
0613          else if ( status .ne. 0 ) then          ! Error reading i
0614          60610          continue          ! Index access error brar
0615          unlock (unit=index_unit)
0616          type *
0617          type *, 'Error accessing MUSIC experiment index, code ', statu
0618          60620          continue          ! Error exit.
0619          type *, 'Eigenvector problem processing abandoned.'
0620          type *
0621          stop
0622          end if
0623          rewrite (unit=index_unit,iostat=status,err=60610)
0624          1          mi_pkey, mi_stat, mi_park, mi_lerr, mi_eerr, mi_se
0625          2          mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_
0626          3          mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wl
0627          4          mi_tol, mi_which, mi_ofile, mi_anolises, mi_irislin
0628          5          mi_zeigs, mi_doas, mi_estype, mi_parp
0629          unlock (unit=index_unit)
0630          go to 10120
0631          c
0632          c          No more status 'l' or 'i' records available.
0633          c
0634          70000          continue
0635          sch_stat = 'l'          ! Revert to main search.
0636          if ( waited .ge. wait_lim ) then          ! Already waited 20 minut
0637              waited = 0
0638              if ( endspawn_if( spawn_unit, my_prog ) ) stop
0639          else
0640              waited = waited + 1
0641              call lib$wait( 300.0 )          ! Wait 5 minutes & try ag
0642          end if
0643          go to 10120
0644          c
0645          c*****
0646          c
0647          c          File error processing branches.
0648          c
0649          c*****
0650          c
0651          c          Computation error, or error writing eigenvalue solutions f

```

EIG_MUSIC

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```

0652      c
0653      60520      continue
0654      close (unit=data_unit)
0655      c
0656      60530      continue
0657      call dump_va( vstatus, msg )
0658      call exit_va( vstatus, msg )
0659      c
0660      60540      continue
0661      mi_stat = 'X'
0662      go to 60600
0663      c
0664      c          Error reading input data file.
0665      c
0666      60500      continue
0667      close (unit=data_unit)
0668      c
0669      60510      continue
0670      call dump_va( vstatus, msg )
0671      call exit_va( vstatus, msg )
0672      c
0673      if ( mi_eerr .ge. 10 ) then
0674      mi_stat = 'X'
0675      else
0676      mi_eerr = mi_eerr + 1
0677      if ( ancestor ) then
0678      mi_stat = ' '
0679      else
0680      mi_stat = 'D'
0681      end if
0682      end if
0683      go to 60600
0684      c
0685      c          Error opening input data file - could be in use by another
0686      c          program.
0687      c
0688      61510      continue
0689      call dump_va( vstatus, msg )
0690      call exit_va( vstatus, msg )
0691      c
0692      if ( status .eq. FOR+IOS_OPEFAL ) then          ! Probably locked.
0693      mi_stat = 'i'
0694      else          ! Some other error.
0695      type *
0696      type *, 'Failed opening input data file ', work_file, ', error #',
0697      1          status, '.'
0698      type *
0699      mi_stat = 'X'          ! Mark for deletion.
0700      end if
0701      go to 60600
0702      c
0703      c          Error performing internal write - abandon efforts.
0704      c
0705      62510      continue
0706      type *
0707      type *, 'Failed performing internal write.'
0708      go to 60620

```

EIG_MUSIC

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```

0709      c
0710      90020      format( i10.10 )
0711      c
0712                      end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	2165	PIC CON REL LCL SHR EXE RD NOWRT
1 \$PDATA	500	PIC CON REL LCL SHR NOEXE RD NOWRT
2 \$LOCAL	1296	PIC CON REL LCL NOSHR NOEXE RD WRT
3 MUSIC_MEASURES	120	PIC OVR REL GBL SHR NOEXE RD WRT
Total Space Allocated	4081	

ENTRY POINTS

Address	Type	Name
0-00000000		EIG_MUSIC

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type	Name
**	L*4	ANCESTOR	2-00000108	R*8	CONDIT	3-0000000C	I*4	CY
2-00000178	I*4	EIGV_ADD	2-00000174	I*4	EIG_ADD	2-00000154	I*4	INI
2-00000118	I*4	MI_ANTS	2-00000140	R*4	MI_BEGP	2-00000148	R*4	MI
2-0000000D	CHAR	MI_DTIM	2-000000DC	I*2	MI_EERR	2-00000144	R*4	MI
2-000000DA	I*2	MI_IERR	2-00000130	I*4	MI_IRLSLIM	2-00000120	I*4	MI
2-00000020	CHAR	MI_OFILE	2-00000114	I*4	MI_PARK	2-0000014C	R*4	MI
2-00000110	I*4	MI_PKEY	2-000000DE	I*2	MI_SERR	2-000000E0	R*8	MI
2-0000000C	CHAR	MI_STAT	2-00000100	R*8	MI_TOL	2-0000011C	I*4	MI
2-000000F8	R*8	MI_WLEN	2-00000134	I*4	MI_ZEIGS	2-0000008A	CHAR	MS
3-00000060	R*8	NON_TRI	3-00000068	I*4	NON_TRI_CYC	2-0000016C	I*4	OB
3-0000006C	R*8	QPSI_LMARG	3-00000074	I*4	QPSI_L_ZCYC	2-00000168	I*4	RB
2-00000070	CHAR	SCH_STAT	3-00000008	I*4	SCORR	2-0000015C	I*4	SP
**	I*4	SP_NUM	**	CHAR	SP_PROC	**	I*4	SP
3-00000010	R*8	SV_INERR	3-00000000	R*8	TOL	3-00000018	R*8	UP
3-00000020	R*8	UPHI_SMARG	3-00000034	I*4	UPHI_S_ZCYC	3-00000028	R*8	U_
3-0000003C	R*8	VPSI_LMARG	3-00000054	I*4	VPSI_L_ZCYC	3-00000044	R*8	VP
2-00000160	I*4	VSTATUS	3-0000004C	R*8	V_UNIERR	3-0000005C	I*4	V_
**	I*4	WAIT_LIM	2-00000071	CHAR	WORK_FILE			

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16-Apr-1988 10:13:54

```

0001      program fin_music
0002      c
0003      c:.....:
0004      c
0005      c          This program lurks in the SYS$SLOWBATCH queue (usually), and
0006      c          pounces on any record with a status of 'D' in MUSIC experiment index
0007      c          file 6000$RT:[CHUCK.PARAMS]MUSIC_INDEX.DAT. When it finds such a
0008      c          record, it reads the information necessary to solve the Lp estimation
0009      c          problem from the appropriate data files, and proceeds to generate the
0010      c          required Lp estimates for the experiment specified in the selected
0011      c          index file record. The program then writes its results to an estimates
0012      c          data file (selected by the experiment number), and updates the MUSIC
0013      c          experiment index record to a status of 'I'. The above sequence is
0014      c          repeated until no more status 'D' records are found AND no process
0015      c          named "Chuck MUSIC Drv" can be found on the VAX.
0016      c          The above description applies exactly to error-free operation
0017      c          only. If a computational error occurs during the estimation process,
0018      c          or a file error occurs during reading of input data or writing of the
0019      c          solutions, the program changes the index record's status to 'X' instead
0020      c          of 'I'. Finally, the program changes the index record's status to
0021      c          'd' if a data file can't be opened (the assumption being that the file
0022      c          is in use by another program). Status 'd' records are processed after
0023      c          all status 'D' record operations have been completed.
0024      c
0025      c:.....:
0026      c
0027      c          implicit none
0028      c          external lib$get_lun, lib$stop, lib$wait, lib$delete_file
0029      c          external lib$init_timer, lib$stat_timer
0030      c          external init_va, open_va, free_va, exit_va, dump_va
0031      c          external readarr, writearr, skiparr
0032      c          external c16_sel_vdiff, sigmat_estm, music_lpp_each
0033      c          external music_lpw_yar, music_est_rep, endspawn_if, spawn_if
0034      c          external accum_doas, mreaddoas, dc_matmpyherm
0035      c          external sqdc_mat_intdiv, music_lpp_speis, music_out_rep
0036      c          external dc_matconjg, dc_mathermmpy, dc_matmpy, c16_vachol, dc_matcopy
0037      c          external bracket_doas, c16_mat_dr_leftmul, dc_enfherm
0038      c
0039      c          include '6000$RT:[CHUCK.RESEARCH.FORTDIR]JVATYPES.TXT'
0051      c          include '6000$RT:[CHUCK.RESEARCH.FORTDIR]MUSIC_MEASURES.FOR'
0140      c          include '6000$RT:[CHUCK.RESEARCH.FORTDIR]MUSIC_INDEX_DEF.TXT'
0281      c          include '6000$RT:[CHUCK.RESEARCH.FORTDIR]SPAWNED_DEF.TXT'
0324      c          include 'SYS$LIBRARY:FORIOSDEF'
0410      c
0411      c          integer*4 my_prog, next_prog
0412      c          parameter (my_prog = 3)
0413      c          parameter (next_prog = 2)
0414      c
0415      c          logical readarr, writearr, skiparr, endspawn_if
0416      c          integer*4 lib$init_timer, lib$stat_timer, lib$get_lun, spawn_if
0417      c
0418      c          logical last, have
0419      c          integer*4 status, index_unit, data_unit2, data_unit, spawn_unit
0420      c          integer*4 waited, wait_lim/ 4 /, vstatus
0421      c          integer*4 tstatus, ptics, gtics
0422      c          integer*4 mnitlim, atitlim
0423      c          integer*4 ants_by_2( 2 ), ants_by_wvs( 2 ), waves_lin( 1 )

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0424 integer*4 smp_by_ants( 2 ), wavs_by_smp( 2 ), ants_lin( 1 )
0425 integer*4 wavs_sqr( 2 ), ants_sqr( 2 )
0426 integer*4 ant_add, doa_add, amat_add
0427 integer*4 obsmat_add, noisemat_add, pvm_add, resm_add, rescj_add
0428 integer*4 yarw_add, plp_add, robs_add, rnoise_add, rb_add
0429 integer*4 obst_add, noiset_add, rescov_add, ap_add, apah_add
0430 integer*4 eigv_add, doaperf_add
0431 integer*4 ai_pkey, rec_num
0432 integer*4 iters, robscop_add, apahcop_add
0433 real*8 min_del, condit, down
0434 real*8 smul, amlim, amul, omul
0435 character*1 sch_stat
0436 character*25 work_file
0437 character*80 msg
0438
0439 c
0440 integer*2 ti_lerr, ti_eerr, ti_serr
0441 integer*4 ti_pkey, ti_park, ti_ants, ti_waves, ti_noises, ti_smp
0442 integer*4 ti_which, ti_anoises, ti_irlslim, ti_zeigs, ti_doas
0443 integer*4 ti_estype
0444 real*4 ti_begp, ti_endp, ti_delp, ti_parp
0445 real*8 ti_sint, ti_peps, ti_norm, ti_wlen, ti_tol
0446 character*1 ti_stat
0447 character*19 ti_dtim
0448 character*80 ti_ofile
0449
0450 c
0451 c          Get logical unit numbers
0452 c
0453 status = lib$get_lun( index_unit )
0454 if (.not. status) call lib$stop( %val( status ) )
0455 status = lib$get_lun( data_unit )
0456 if (.not. status) call lib$stop( %val( status ) )
0457 status = lib$get_lun( data_unit2 )
0458 if (.not. status) call lib$stop( %val( status ) )
0459 status = lib$get_lun( spawn_unit )
0460 if (.not. status) call lib$stop( %val( status ) )
0461
0462 c
0463 c          Open the MUSIC index file.
0464 c
0465 c          10000 continue
0466 open (unit=index_unit,file='6000*RT:[CHUCK.PARAMS]MUSIC_INDEX.DAT',
0467      1 form='UNFORMATTED',recordtype='FIXED',recl=64,
0468      2 organization='INDEXED',access='KEYED',status='OLD',
0469      3 key=(1:4:INTEGER,5:5:CHARACTER,6:9:INTEGER),dispose='KEEP',
0470      4 SHARED,err=10100)
0471 go to 10110
0472
0473 c
0474 c          Error opening the MUSIC index file or MUSIC spawned-processes:
0475 c          index file, abandon hope.
0476 c
0477 c          10100 continue
0478 type *
0479 type *, 'Error opening MUSIC index file or spawned-processes file.
0480 go to 60620
0481
0482 c
0483 c          MUSIC index file successfully opened. Open MUSIC spawned-
0484 c          processes index file.
0485 c
0486 c

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0481 20110 continue
0482 open (unit=spawn_unit,file='6000*RT:[CHUCK.PARAMS]SPAWNED.IDX',
0483 1 form='UNFORMATTED',recordtype='FIXED',recl=26,
0484 2 organization='INDEXED',access='KEYED',status='OLD',
0485 3 key=(1:4:INTEGER,5:8:INTEGER),dispose='KEEP',SHARED,err=10100)
0486 c
0487 c MUSIC index file and MUSIC spawned-processes index file
0488 c successfully opened. Start lurking, waiting to pounce on unsuspecting
0489 c status 'D' records.
0490 c
0491 c waited = 0
0492 c sch_stat = 'D'
0493 20120 continue
0494 read (unit=index_unit,keyeq=sch_stat,keyid=1,iostat=status,err=20100)
0495 1 mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
0496 2 mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
0497 3 mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
0498 4 mi_tol, mi_which, mi_ofile, mi_anaises, mi_irislim,
0499 5 mi_zeigs, mi_doas, mi_estype, mi_parp
0500 unlock (unit=index_unit)
0501 sch_stat = 'D'
0502 waited = 0
0503 go to 20110
0504 c
0505 c Error reading status 'D' records. Could be that the first such
0506 c record is locked, no such records exist, or some other error.
0507 c
0508 20100 continue
0509 unlock (unit=index_unit)
0510 if ( status .eq. FOR%IOS_SPERECLOC ) then ! Locked.
0511 type *, 'Record locked, status ', sch_stat, '. Waiting 5 min.'
0512 waited = 0
0513 20120 continue ! Branch to wait then read.
0514 call lib$wait( 300.0 ) ! Hibernate for 5 minutes.
0515 go to 10120
0516 else if ( status .eq. FOR%IOS_ATTACCNON ) then ! No record.
0517 if ( sch_stat .eq. 'D' ) then ! May be 'd's.
0518 sch_stat = 'd'
0519 go to 10120
0520 else ! Neither 'D's nor 'd's.
0521 go to 70000
0522 end if
0523 else ! Some other error.
0524 60830 continue
0525 type *
0526 type *, 'Failed reading index records or accessing scratch file.'
0527 call lib$stop( %val( status ) )
0528 end if
0529 20110 continue
0530 c
0531 c Successfully read a status 'D' (or 'd') record into the batch
0532 c of variables listed in the read statement above. Now determine the
0533 c current experiment's ancestor's key number. Also collect DOA estimates
0534 c to scratch disk file.
0535 c
0536 c call init_va
0537 open (unit=data_unit2,file='6000*RT:[CHUCK.PARAMS]FMSCRATCH.DAT',

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0538      1      form='UNFORMATTED',recordtype='FIXED',rec1=2,
0539      2      organization='RELATIVE',access='DIRECT',
0540      3      status='NEW',dispose='DELETE',iostat=status,err=60830)
0541      ti_doas = 0
0542      write (unit=data_unit2,rec=1,iostat=status,err=60830) ti_doas
0543      ai_pkey = mi_pkey
0544      ti_doas = mi_doas
0545      ti_park = mi_park
0546      work_file = 'MUSIC_SAMPL_____DAT'
0547
0548      c
0549      do while ( ti_park .ne. -1 ) ! Until we find the "ancestor".
0550      write (unit=work_file( 12 : 21 ),fmt=90020,err=60510) ai_pkey
0551      open (unit=data_unit,file='6000$RT:[CHUCK.PARAMS]//work_file,
0552      1      status='OLD',organization='SEQUENTIAL',access='SEQUENTIAL',
0553      2      form='UNFORMATTED',recordtype='FIXED',rec1=64,dispose='KEEP',
0554      3      iostat=status,READONLY,err=61500)
0555
0556      c
0557      waves_lin(1) = ti_doas
0558      call open_va( doa_add, waves_lin, 1, dr_type, vstatus, msg )
0559      call exit_va( vstatus, msg )
0560      if ( .not. ( readarr( Xval( doa_add ), dr_type, ti_doas, 1,
0561      1      data_unit, 32 ) ) ) go to 60500
0562      close (unit=data_unit)
0563
0564      c
0565      call accum_doas( ti_doas, Xval( doa_add ), data_unit2 )
0566      call free_va( doa_add, vstatus, msg )
0567      call exit_va( vstatus, msg )
0568
0569      c
0570      ai_pkey = ti_park
0571      21120      continue
0572      read (unit=index_unit,keyeq=ai_pkey,keyid=0,iostat=status,
0573      1      err=21100)
0574      2      ai_pkey, ti_stat, ti_park, ti_ierr, ti_eerr, ti_serr,
0575      3      ti_dtim, ti_ants, ti_waves, ti_noises, ti_smp, ti_sint,
0576      4      ti_begp, ti_endp, ti_delp, ti_peps, ti_norm, ti_wlen,
0577      5      ti_tol, ti_which, ti_ofile, ti_anoises, ti_lrlslim,
0578      6      ti_zeigs, ti_doas, ti_estype, ti_parp
0579      unlock (unit=index_unit)
0580      go to 21110
0581
0582      c
0583      Error reading status 'D' record ancestor chain. Could be that
0584      c      a record is locked, or some other error.
0585      c
0586      21100      continue
0587      unlock (unit=index_unit)
0588      if ( status .eq. FOR$IOS_SPERECLOC ) then ! Locked.
0589      type *, 'Record #', ai_pkey, ' locked. Waiting 5 min.'
0590      21130      continue ! Branch to wait then read.
0591      call lib$wait( 300.0 ) ! Hibernate for 5 minutes.
0592      go to 21120
0593      else ! Some other error.
0594      go to 60830 ! Abandon program.
0595      end if
0596
0597      c
0598      21110      continue ! Successful read.
0599      end do ! End of loop to find "ancestor".
0600
0601      c

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0595 c           Get accumulated DOA estimates from scratch file, then close
0596 c           (and, as a result, delete) the file.
0597 c
0598 read (unit=data_unit2,rec=1) waves_lin(1)
0599 call open_va( doa_add, waves_lin, 1, dr_type, vstatus, msg )
0600 call exit_va( vstatus, msg )
0601 call open_va( doaperf_add, waves_lin, 1, dr_type, vstatus, msg )
0602 call exit_va( vstatus, msg )
0603 rec_num = 2
0604 call mreaddoas( data_unit2, rec_num, %val( doa_add ), waves_lin(1) )
0605 close (unit=data_unit2)
0606 c
0607 c           Commence estimation process.
0608 c
0609 wavs_sqr(1) = waves_lin(1)
0610 wavs_sqr(2) = waves_lin(1)
0611 ants_lin(1) = mi_ants
0612 ants_sqr(1) = mi_ants
0613 ants_sqr(2) = mi_ants
0614 ants_by_wvs(1) = mi_ants
0615 ants_by_wvs(2) = waves_lin(1)
0616 ants_by_2(1) = mi_ants
0617 ants_by_2(2) = 2
0618 wavs_by_smp(1) = waves_lin(1)
0619 wavs_by_smp(2) = mi_smp
0620 smp_by_ants(1) = mi_smp
0621 smp_by_ants(2) = mi_ants
0622 c
0623 call open_va( ant_add, ants_by_2, 2, dr_type, vstatus, msg )
0624 call exit_va( vstatus, msg )
0625 call open_va( amat_add, ants_by_wvs, 2, dc_type, vstatus, msg )
0626 call exit_va( vstatus, msg )
0627 call open_va( obsmat_add, smp_by_ants, 2, dc_type, vstatus, msg )
0628 call exit_va( vstatus, msg )
0629 call open_va( pvm_add, wavs_by_smp, 2, dc_type, vstatus, msg )
0630 call exit_va( vstatus, msg )
0631 call open_va( resm_add, smp_by_ants, 2, dc_type, vstatus, msg )
0632 call exit_va( vstatus, msg )
0633 call open_va( rescj_add, smp_by_ants, 2, dc_type, vstatus, msg )
0634 call exit_va( vstatus, msg )
0635 call open_va( plp_add, wavs_sqr, 2, dc_type, vstatus, msg )
0636 call exit_va( vstatus, msg )
0637 call open_va( robs_add, ants_sqr, 2, dc_type, vstatus, msg )
0638 call exit_va( vstatus, msg )
0639 call open_va( rescov_add, ants_sqr, 2, dc_type, vstatus, msg )
0640 call exit_va( vstatus, msg )
0641 call open_va( rb_add, ants_sqr, 2, dc_type, vstatus, msg )
0642 call exit_va( vstatus, msg )
0643 call open_va( obst_add, ants_sqr, 2, dc_type, vstatus, msg )
0644 call exit_va( vstatus, msg )
0645 call open_va( noiset_add, ants_sqr, 2, dc_type, vstatus, msg )
0646 call exit_va( vstatus, msg )
0647 call open_va( eigv_add, ants_sqr, 2, dc_type, vstatus, msg )
0648 call exit_va( vstatus, msg )
0649 call open_va( ap_add, ants_by_wvs, 2, dc_type, vstatus, msg )
0650 call exit_va( vstatus, msg )
0651 call open_va( apah_add, ants_sqr, 2, dc_type, vstatus, msg )

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0652      call exit_va( vstatus, msg )
0653      call open_va( robscop_add, ants_sqr, 2, dc_type, vstatus, msg )
0654      call exit_va( vstatus, msg )
0655      call open_va( apahcop_add, ants_sqr, 2, dc_type, vstatus, msg )
0656      call exit_va( vstatus, msg )
0657      if ( ( mi_estype .ne. 1 ) .and. ( mi_estype .ne. 3 ) ) then ! SPwr.
0658          call open_va( yarw_add, ants_lin, 1, dr_type, vstatus, msg )
0659          call exit_va( vstatus, msg )
0660          call open_va( rnoise_add, ants_sqr, 2, dc_type, vstatus, msg )
0661          call exit_va( vstatus, msg )
0662      end if
0663      if ( ( mi_estype .eq. 3 )
0664          .or. ( mi_estype .eq. 5 )
0665          .or. ( mi_estype .eq. 6 ) ) then ! Compensated estim.
0666          call open_va( noisemat_add, smp_by_ants, 2, dc_type, vstatus, msg )
0667          call exit_va( vstatus, msg )
0668      end if
0669      c
0670      c          Get observation (and, for "compensated" estimation types,
0671      c          noise) samples from the "ancestor" experiment's "samples" file.
0672      c
0673      write (unit=work_file( 12 : 21 ),fmt=90020,err=60530) ai_pkey
0674      open (unit=data_unit,file='6000$RT:[CHUCK.PARAMS]//work_file,
0675      1      status='OLD',organization='SEQUENTIAL',access='SEQUENTIAL',
0676      2      form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
0677      3      iostat=status,READONLY,err=61510)
0678      if ( .not. ( readarr( %val( obsmat_add ), dc_type, mi_smp, mi_ants,
0679      1      data_unit, 16 ) ) ) go to 60520
0680      if ( ( mi_estype .eq. 3 )
0681          .or. ( mi_estype .eq. 5 )
0682          .or. ( mi_estype .eq. 6 ) ) then
0683          if ( .not. ( readarr( %val( noisemat_add ), dc_type, mi_smp,
0684      1      mi_ants, data_unit, 16 ) ) ) go to 60520
0685          call c16_sel_vdiff( %val( obsmat_add ), 1,
0686      1      %val( noisemat_add ), 1, mi_smp * mi_ants,
0687      2      %val( obsmat_add ) )
0688      end if
0689      close (unit=data_unit)
0690      c
0691      c          Get antenna array definition from the "ancestor" experiment's
0692      c          "input" file.
0693      c
0694      work_file( 7 : 11 ) = 'INPUT'
0695      open (unit=data_unit,file='6000$RT:[CHUCK.PARAMS]//work_file,
0696      1      status='OLD',organization='SEQUENTIAL',access='SEQUENTIAL',
0697      2      form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
0698      3      iostat=status,READONLY,err=61510)
0699      read (unit=data_unit,err=60520)
0700      if ( .not. ( readarr( %val( ant_add ), dr_type, mi_ants, 2,
0701      1      data_unit, 32 ) ) ) go to 60520
0702      close (unit=data_unit)
0703      c
0704      c          Get observation, noise (for "sum of power" estimation types),
0705      c          and normalized noise covariance matrices, and QR decomposition result
0706      c          matrices from the "ancestor" experiment's "eigenvectors" file.
0707      c
0708      c          Also get eigenvectors.

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0709 c
0710 work_file( 7 : 11 ) = 'EIGVS'
0711 open (unit=data_unit,file='6000$RT:[CHUCK.PARAMS]//work_file,
0712 1 status='OLD',organization='SEQUENTIAL',access='SEQUENTIAL',
0713 2 form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
0714 3 iostat=status,READONLY,err=61510)
0715 if ( .not. ( readarr( %val( robs_add ), dc_type, mi_ants, mi_ants,
0716 1 data_unit, 16 ) ) ) go to 60520
0717 if ( ( mi_estype .ne. 1 ) .and. ( mi_estype .ne. 3 ) ) then
0718 if ( .not. ( readarr( %val( rnoise_add ), dc_type, mi_ants,
0719 1 mi_ants, data_unit, 16 ) ) ) go to 60520
0720 else
0721 if ( .not. ( skiparr( mi_ants, mi_ants, data_unit, 16 ) ) )
0722 1 go to 60520
0723 end if
0724 if ( ti_which .gt. 0 ) then
0725 if ( .not. ( skiparr( mi_ants, mi_ants, data_unit, 16 ) ) )
0726 1 go to 60520
0727 end if
0728 if ( .not. ( readarr( %val( rb_add ), dc_type, mi_ants, mi_ants,
0729 1 data_unit, 16 ) ) ) go to 60520
0730 if ( .not. ( skiparr( mi_ants, mi_ants, data_unit, 16 ) ) )
0731 1 go to 60520
0732 if ( .not. ( readarr( %val( noiset_add ), dc_type, mi_ants, mi_ants,
0733 1 data_unit, 16 ) ) ) go to 60520
0734 if ( .not. ( skiparr( mi_ants, 1, data_unit, 32 ) ) ) go to 60520
0735 if ( .not. ( readarr( %val( eigv_add ), dc_type, mi_ants, mi_ants,
0736 1 data_unit, 16 ) ) ) go to 60520
0737 close (unit=data_unit)
0738 c
0739 c Get DDA estimate "peakedness" measures.
0740 c
0741 down = 1.0d0
0742 call bracket_doas( %val( doa_add ), waves_lin(1), %val( ant_add ),
0743 1 mi_ants, mi_wlen, %val( eigv_add ), mi_zeigs,
0744 2 mi_parp, %val( doaperf_add ), down )
0745 c
0746 c Form estimate of A matrix.
0747 c
0748 call sigmat_estm( %val( ant_add ), mi_ants, %val( doa_add ),
0749 1 waves_lin(1), mi_wlen, %val( amat_add ) )
0750 c
0751 c Form estimates of signal and residual vectors and P matrix as
0752 c selected by the estimation type specified for the experiment.
0753 c
0754 tstatus = lib$init_timer()
0755 if ( .not. tstatus ) call lib$stop( %val( tstatus ) )
0756 if ( ( mi_estype .eq. 1 ) ! Enumeration.
0757 1 .or. ( mi_estype .eq. 3 ) ) then ! Comp'd enum.
0758 call music_1pp_each( %val( obsmat_add ), mi_smp, mi_ants,
0759 1 %val( amat_add ), waves_lin(1), mi_parp,
0760 2 mi_peps, mi_norm, mi_tol, %val( pvm_add ),
0761 3 %val( resm_add ), mi_irlslim, mnitlim,
0762 4 atitlim, min_del )
0763 call dc_matmpyherm( %val( pvm_add ), waves_lin(1), mi_smp,
0764 1 %val( pvm_add ), waves_lin(1),
0765 2 %val( plp_add ) )

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0766      call sqdc_mat_intdiv( %val( plp_add ), waves_lin(1), mi_smp )
0767      else if ( ( mi_estype .eq. 2 ) ) then ! Sum powers.
0768      1      .or. ( mi_estype .eq. 5 ) ) then ! C'd sum pwr.
0769      call music_lpw_yar( %val( obsmat_add ), mi_smp, mi_ants,
0770      1      %val( amat_add ), waves_lin(1), mi_parp,
0771      2      mi_peps, mi_norm, mi_tol, %val( yarw_add ),
0772      3      %val( pvm_add ), %val( resm_add ),
0773      4      mi_lrslim, mnitlim, min_del, 1 )
0774      call music_lpp_speis( %val( amat_add ), mi_ants, waves_lin(1),
0775      1      %val( yarw_add ), %val( robs_add ),
0776      2      %val( rnoise_add ), mi_tol,
0777      3      %val( plp_add ), status )
0778      if ( status .ne. 0 ) mnitlim = - mnitlim - 1
0779      else if ( ( mi_estype .eq. 4 ) ) then ! S. pwr loss.
0780      1      .or. ( mi_estype .eq. 6 ) ) then ! C. SPw loss.
0781      call music_lpw_yar( %val( obsmat_add ), mi_smp, mi_ants,
0782      1      %val( amat_add ), waves_lin(1), mi_parp,
0783      2      mi_peps, mi_norm, mi_tol, %val( yarw_add ),
0784      3      %val( pvm_add ), %val( resm_add ),
0785      4      mi_lrslim, mnitlim, min_del, 2 )
0786      call music_lpp_speis( %val( amat_add ), mi_ants, waves_lin(1),
0787      1      %val( yarw_add ), %val( robs_add ),
0788      2      %val( rnoise_add ), mi_tol,
0789      3      %val( plp_add ), status )
0790      if ( status .ne. 0 ) mnitlim = - mnitlim - 1
0791      else ! Invalid method.
0792      go to 60530 ! Mark for deletion.
0793      end if
0794      c
0795      c      Compute residuals covariance matrix, updated observation
0796      c      covariance matrix, and Cholesky decomposition of updated covariance
0797      c      matrix.
0798      c
0799      tstatus = lib$stat_timer( 2, ptics )
0800      if ( .not. tstatus ) call lib$stop( %val( tstatus ) )
0801      tstatus = lib$inlt_timer()
0802      if ( .not. tstatus ) call lib$stop( %val( tstatus ) )
0803      c
0804      call dc_matcopy( %val( resm_add ), mi_smp, mi_ants,
0805      1      %val( rescj_add ) )
0806      call dc_matconjg( %val( rescj_add ), mi_smp, mi_ants )
0807      call dc_mathermmpy( %val( rescj_add ), mi_smp, mi_ants,
0808      1      %val( rescj_add ), mi_ants, %val( rescov_add ) )
0809      call sqdc_mat_intdiv( %val( rescov_add ), mi_ants, mi_smp )
0810
0811      call dc_matmpy( %val( amat_add ), mi_ants, waves_lin(1),
0812      1      %val( plp_add ), waves_lin(1), %val( ap_add ) )
0813      call dc_matmpyherm( %val( ap_add ), mi_ants, waves_lin(1),
0814      1      %val( amat_add ), mi_ants, %val( apah_add ) )
0815      c
0816      c      Determine "almost optimum" multiplier for A P A-conj-transp.
0817      c
0818      iters = 0
0819      smul = 1.0d0
0820      have = .false.
0821      last = .false.
0822      40000      continue

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0823      call dc_matcopy( %val( robs_add ), mi_ants, mi_ants,
0824      1      %val( robscop_add ) )
0825      call dc_matcopy( %val( apah_add ), mi_ants, mi_ants,
0826      1      %val( apahcop_add ) )
0827      call c16_mat_dr_leftmul( %val( apahcop_add ), mi_ants,
0828      1      mi_ants, smul, mi_ants )
0829      call c16_sel_vdiff( %val( robscop_add ), 1, %val( apahcop_add ),
0830      1      1, mi_ants * mi_ants, %val( robscop_add ) )
0831      call dc_enfherm( %val( robscop_add ), mi_ants, mi_tol )
0832      call c16_vachol( %val( robscop_add ), mi_ants,
0833      1      %val( obst_add ), condit, status )
0834      if ( last ) go to 40010      ! Just re-did last working.
0835      if ( have ) then          ! Have a working multiplier.
0836      if ( status .eq. 0 ) then ! Cholesky succeeded.
0837      omul = smul
0838      else                      ! Cholesky failed.
0839      smul = omul
0840      end if
0841      if ( amul .ge. amlim ) then ! Insufficient resolution.
0842      amul = amul * 0.5d0      ! Up the resolution some.
0843      smul = smul + amul      ! Try slightly larger mult.
0844      else                    ! Sufficient resolution.
0845      last = .true.          ! Set last time through.
0846      end if
0847      else
0848      if ( status .eq. 0 ) then ! Don't have working yet.
0849      ! Cholesky succeeded.
0850      amlim = smul * 0.0078125d0 ! Set resolution limit.
0851      amul = smul * 0.5d0      ! Set first mult. add term.
0852      omul = smul            ! Save working multiplier.
0853      smul = smul + amul      ! Try slightly larger mult.
0854      have = .true.          ! Indicate we have working m.
0855      else                  ! Cholesky failed.
0856      ! Indicate another pass.
0857      if ( iters .gt. 25 ) then ! Tried all we care to.
0858      smul = 0.0d0          ! Don't subtract any.
0859      last = .true.          ! Indicate last attempt.
0860      else                  ! Haven't tried too many.
0861      smul = smul * 0.5d0    ! Try half of previous mult.
0862      end if
0863      end if
0864      go to 40000            ! Try again.
0865      40010      continue
0866      c
0867      tstatus = lib$stat_timer( 2, gtics )
0868      if ( .not. tstatus ) call lib$stop( %val( tstatus ) )
0869      call dc_matcopy( %val( robscop_add ), mi_ants, mi_ants,
0870      1      %val( robs_add ) )
0871      c
0872      c      Print estimation results, and cancel further processing if a
0873      c      computation failure has occurred.
0874      c
0875      call music_est_rep( data_unit, mi_ofile, mi_dtim, mi_pkey,
0876      1      mi_park, %val( doa_add ), %val( doaperf_add ),
0877      2      %val( pip_add ), waves_lin(1), smul, down,
0878      3      %val( amat_add ), mi_ants, mi_parp, mi_estype,
0879      4      mi_irislim, mnitlim, atitlim, min_del, condit,

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0880      5          status, ptics / 100.0d0, gtics / 100.0d0 )
0881      call music_out_rep( data_unit, mi_pkey, mi_park, mi_parp, ai_pkey,
0882      1          mi_which, mi_ofile, %val( robs_add ),
0883      2          0, 0, mi_ants, 0.0d0, mi_dtim )
0884      if ( ( status .ne. 0 ) .or. ( mntlim .lt. 0 ) ) go to 60530
0885      c
0886      c          Write signal and residual vector estimates to output file.
0887      c
0888      work_file( 7 : 11 ) = 'SAMPL'
0889      write (unit=work_file( 12 : 21 ),fmt=90020,err=60530) mi_pkey
0890      open (unit=data_unit,file='6000*RT:[CHUCK.PARAMS]//work_file,
0891      1          status='OLD',organization='SEQUENTIAL',
0892      2          access='SEQUENTIAL',form='UNFORMATTED',
0893      3          recordtype='FIXED',recl=64,dispose='KEEP',err=60530)
0894      if ( .not. ( skiparr( mi_dcoas, 1, data_unit, 32 ) ) ) go to 60520
0895      if ( .not. ( writearr( %val( resm_add ), dc_type, mi_smp,
0896      1          mi_ants, data_unit, 16 ) ) ) go to 60520
0897      if ( .not. ( writearr( %val( pvm_add ), dc_type, waves_lin(1),
0898      1          mi_smp, data_unit, 16 ) ) ) go to 60520
0899      close (unit=data_unit)
0900      c
0901      c          Create eigenvectors file for the current experiment, and write
0902      c          the covariance and factored covariance matrices to the file.
0903      c
0904      work_file( 7 : 11 ) = 'EIGVS'
0905      call lib$delete_file(
0906      1          '6000*RT:[CHUCK.PARAMS] // work_file // ',*')
0907      open (unit=data_unit,file='6000*RT:[CHUCK.PARAMS]//work_file,
0908      1          status='NEW',organization='SEQUENTIAL',access='SEQUENTIAL',
0909      2          form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
0910      3          err=60530)
0911      if ( .not. ( writearr( %val( rescov_add ), dc_type, mi_ants,
0912      1          mi_ants, data_unit, 16 ) ) ) go to 60520
0913      if ( .not. ( writearr( %val( robs_add ), dc_type, mi_ants,
0914      1          mi_ants, data_unit, 16 ) ) ) go to 60520
0915      if ( .not. ( writearr( %val( rb_add ), dc_type, mi_ants,
0916      1          mi_ants, data_unit, 16 ) ) ) go to 60520
0917      if ( .not. ( writearr( %val( obst_add ), dc_type, mi_ants,
0918      1          mi_ants, data_unit, 16 ) ) ) go to 60520
0919      if ( .not. ( writearr( %val( noiset_add ), dc_type, mi_ants,
0920      1          mi_ants, data_unit, 16 ) ) ) go to 60520
0921      close (unit=data_unit)
0922      c
0923      c          Discard space used by Virtual Arrays, update the experiment's
0924      c          index record, and exit if there is no more work to do.
0925      c
0926      call dump_va( vstatus, msg )
0927      call exit_va( vstatus, msg )
0928      mi_stat = 'I'
0929      c
0930      status = spawn_if( spawn_unit, next_prog )      ! Start eigenvalues.
0931      if ( status .ne. 0 ) then                        ! Spawn failed.
0932      type *,
0933      type *, 'Error initiating eigenvalue solver.'
0934      if ( status .eq. 1 ) then
0935      type *, 'Read error on spawned-processes index file.'
0936      else if ( status .eq. 2 ) then

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0937         type *, 'Error spawning subprocess.'
0938     else if ( status .eq. 3 ) then
0939         type *, 'Error rewriting spawned-processes index file.'
0940     else
0941         type *, 'Impossible error, code ', status, ' '
0942     end if
0943     go to 60620
0944 end if
0945 c
0946 60600 continue
0947 read (unit=index_unit,key=mi_pkey,keyid=0,iostat=status)
0948 if ( status .eq. FOR$IOS_SPERECLC ) then
0949     type *, 'Locked record to rewrite, #', mi_pkey,
0950     1     ' . Waiting 10 sec.'
0951     unlock (unit=index_unit)
0952     call lib$wait( 10.0 ) ! Wait 10 seconds.
0953     go to 60600
0954 else if ( status .ne. 0 ) then ! Error reading index.
0955     60610 continue ! Index access error branch.
0956     unlock (unit=index_unit)
0957     type *
0958     type *, 'Error accessing MUSIC experiment index, code ', status
0959     60620 continue ! Error exit.
0960     type *, 'Lp estimation abandoned.'
0961     type *
0962     stop
0963 end if
0964 rewrite (unit=index_unit,iostat=status,err=60610)
0965 1     mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
0966 2     mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
0967 3     mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
0968 4     mi_tol, mi_which, mi_ofile, mi_anaises, mi_lrislim,
0969 5     mi_zeigs, mi_doas, mi_estype, mi_parp
0970 unlock (unit=index_unit)
0971 go to 10120
0972 c
0973 c     No more status 'D' records available.
0974 c
0975 70000 continue
0976 if ( waited .ge. wait_lim ) then ! Already waited 20 minutes.
0977     waited = 0
0978     if ( endspawn_if( spawn_unit, my_prog ) ) stop
0979 else
0980     waited = waited + 1
0981     call lib$wait( 300.0 ) ! Wait 5 minutes & try again.
0982 end if
0983 go to 10120
0984 c
0985 c*****
0986 c
0987 c     File error processing branches.
0988 c
0989 c*****
0990 c
0991 c     Computation or file access error.
0992 c
0993 60500 continue

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0994          close (unit=data_unit)
0995      c
0996      60510  continue
0997          close (unit=data_unit2)
0998          go to 60530
0999      c
1000      60520  continue
1001          close (unit=data_unit)
1002      c
1003      60530  continue
1004          call dump_va( vstatus, msg )
1005          call exit_va( vstatus, msg )
1006      c
1007      60540  continue
1008          mi_stat = 'X'
1009          go to 60600
1010      c
1011      c          Error opening input data file - could be in use by another
1012      c          program.
1013      c
1014      61500  continue
1015          close (unit=data_unit2)
1016      c
1017      61510  continue
1018          call dump_va( vstatus, msg )
1019          call exit_va( vstatus, msg )
1020      c
1021          if ( status .eq. FOR$IOS_OPEFAI ) then          ! Probably locked.
1022              mi_stat = 'd'
1023          else          ! Some other error.
1024              type *
1025              type *, 'Failed opening input data file ', work_file, ', error #',
1026              1          status, '.'
1027              type *
1028              mi_stat = 'X'          ! Mark for deletion.
1029          end if
1030          go to 60600
1031      c
1032      90020  format( i10.10 )
1033      c
1034          end

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```

0001      program music_pest
0002      c
0003      c::::::::::::::::::::::::::::::::::::::::::::::::::::::::::::::::::
0004      c
0005      c          This program lurks in the SYS$SLOWBATCH queue (usually), and
0006      c          pounces on any file matching the specification
0007      c          6000$RT:[CHUCK.PARAMS]MUSIC_STDOAXXXXXXXXXX.DAT. When it finds such a
0008      c          file, it reads the information necessary to solve the Lp estimation
0009      c          problem from the appropriate data files, and proceeds to generate the
0010      c          required Lp estimates of the P matrix for the experiment which
0011      c          created the file. The program then writes its results to the
0012      c          appropriate output file (selected by the experiment number), and
0013      c          deletes the input file. The above sequence is repeated until no more
0014      c          input files are found AND no process named "Chuck MUSIC Drv" can be
0015      c          found on the VAX.
0016      c
0017      c::::::::::::::::::::::::::::::::::::::::::::::::::::::::::::::::::
0018      c
0019      c          implicit none
0020      c          external lib$get_lun, lib$stop, lib$wait
0021      c          external lib$init_timer, lib$stat_timer
0022      c          external lib$find_file, lib$find_file_end, lib$delete_file
0023      c          external init_va, open_va, exit_va, dump_va
0024      c          external endspawn_if, readarr, skiparr
0025      c          external c16_sel_vdiff, sigmat_estm, dc_matmpyherm, sqdc_mat_intdiv
0026      c          external music_l2p, music_lpp_each, music_lpw_yar, music_lpp_speis
0027      c          external music_p_rep
0028      c
0029      c          logical endspawn_if, readarr, skiparr
0030      c          integer*4 lib$init_timer, lib$stat_timer
0031      c          integer*4 lib$get_lun, lib$find_file, lib$find_file_end
0032      c
0033      c          include '6000$RT:[CHUCK.RESEARCH.FORTDIR]VATYPES.TXT'
0034      c          include '6000$RT:[CHUCK.RESEARCH.FORTDIR]MUSIC_MEASURES.FOR'
0035      c          include '6000$RT:[CHUCK.RESEARCH.FORTDIR]MUSIC_INDEX_DEF.TXT'
0036      c          include '6000$RT:[CHUCK.RESEARCH.FORTDIR]SPAWNED_DEF.TXT'
0037      c          include 'SYS$LIBRARY:FORIOSDEF'
0038      c
0039      c          integer*4 my_prog
0040      c          parameter (my_prog = 4)
0041      c
0042      c          logical found
0043      c          integer*4 index_unit, data_unit1, data_unit2, spawn_unit
0044      c          integer*4 status, vstatus, l2status, findadd, waits, waits_lim/ 4 /
0045      c          integer*4 fails, fails_lim/ 100 /
0046      c          integer*4 parent, experiment, ancestor, ants, waves, exps, estim
0047      c          integer*4 waves_lin( 1 ), doa_add, expnum_add
0048      c          integer*4 waves_by_9( 2 ), waves_sqr( 2 ), ants_lin( 1 )
0049      c          integer*4 ants_by_wvs( 2 ), ants_sqr( 2 ), ants_by_2( 2 )
0050      c          integer*4 wavs_by_smp( 2 ), smp_by_ants( 2 )
0051      c          integer*4 wave_add, ant_add, amat_add, obsmat_add, noisemat_add
0052      c          integer*4 pvm_add, resm_add, pl2_add, plp_add, pmat_add, sigvm_add
0053      c          integer*4 robs_add, rnoise_add, yarw_add
0054      c          integer*4 mnitlim, atitlim, tstatus, l2tics, lptics
0055      c          real*4 p
0056      c          real*8 min_del
0057      c          character*25 work_file

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0424      character*80 msg
0425      character*255 found_file
0426      c
0427      c          Get logical unit numbers
0428      c
0429      status = lib$get_lun( index_unit )
0430      if (.not. status) call lib$stop( %val( status ) )
0431      status = lib$get_lun( data_unit1 )
0432      if (.not. status) call lib$stop( %val( status ) )
0433      status = lib$get_lun( data_unit2 )
0434      if (.not. status) call lib$stop( %val( status ) )
0435      status = lib$get_lun( spawn_unit )
0436      if (.not. status) call lib$stop( %val( status ) )
0437      c
0438      c          Open the MUSIC index file.
0439      c
0440      10000  continue
0441      open (unit=index_unit,file='6000*RT:[CHUCK.PARAMS]MUSIC_INDEX.DAT',
0442            1      form='UNFORMATTED',recordtype='FIXED',recl=64,
0443            2      organization='INDEXED',access='KEYED',status='OLD',
0444            3      key=(1:4:INTEGER,5:5:CHARACTER,6:9:INTEGER),dispose='KEEP',
0445            4      SHARED,err=10100)
0446      go to 10110
0447      c
0448      c          Error opening the MUSIC index file or MUSIC spawned-processes
0449      c          index file; abandon hope.
0450      c
0451      10100  continue
0452      type *
0453      type *, 'Error opening MUSIC index file or spawned-processes file.'
0454      c
0455      60620  continue          ! Error exit.
0456      type *, 'Lp estimation abandoned.'
0457      type *
0458      stop
0459      c
0460      c          MUSIC index file successfully opened. Open MUSIC spawned-
0461      c          processes index file.
0462      c
0463      10110  continue
0464      open (unit=spawn_unit,file='6000*RT:[CHUCK.PARAMS]SPAWNED.IDX',
0465            1      form='UNFORMATTED',recordtype='FIXED',recl=26,
0466            2      organization='INDEXED',access='KEYED',status='OLD',
0467            3      key=(1:4:INTEGER,5:8:INTEGER),dispose='KEEP',SHARED,err=10100)
0468      c
0469      c          MUSIC index file and MUSIC spawned-processes index file
0470      c          successfully opened. Start lurking, waiting to pounce on unsuspecting
0471      c          input data files.
0472      c
0473      waits = 0
0474      fails = 0
0475      20000  continue
0476      findadd = 0
0477      found = .false.
0478      21000  continue
0479      status = lib$find_file(
0480      1      '6000*RT:[CHUCK.PARAMS]MUSIC_STD0A%XXXXXXXXX.DAT;*',

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0481          2                                found_file, findadd )
0482          if ( .not. status ) then          ! No input files.
0483          status = lib$find_file_end( findadd ) ! Complete file search.
0484          if ( .not. found ) then          ! Found no files before.
0485          if ( waits .ge. waits_lim ) then ! Already waited 20 minutes.
0486          waits = 0
0487          if ( endspawn_if( spawn_unit, my_prog ) ) stop
0488          else
0489          waits = waits + 1
0490          call lib$wait( 300.0 )          ! Wait 5 minutes & try again.
0491          end if
0492          end if
0493          go to 20000          ! Start new file search.
0494          end if
0495          c
0496          c          Found an input file. Get experiment number, ancestor number,
0497          c          number of experiments in ancestor chain, estimation type to use, and
0498          c          value of p to use for estimation.
0499          c
0500          found = .true.
0501          waits = 0
0502          23000 continue
0503          open (unit=data_unit1,file=found_file,status='OLD',
0504                1 organization='SEQUENTIAL',access='SEQUENTIAL',
0505                2 form='UNFORMATTED',recordtype='FIXED',recl=3,dispose='KEEP',
0506                3 READONLY,iostat=status,err=80000)
0507          go to 80010
0508          c
0509          80000 continue
0510          if ( waits .ge. waits_lim ) then ! Already tried for 20 min.
0511          type *,
0512          type *, 'Failed opening input file:'
0513          type *, found_file
0514          type *, waits_lim, ' times in a row.'
0515          go to 60620
0516          else
0517          call lib$wait( 300.0 )          ! Haven't tried for 20 min.
0518          go to 23000          ! Wait 5 minutes & try again.
0519          end if
0520          c
0521          80010 continue
0522          call init_va          ! Prepare for VA operations.
0523          waits = 0
0524          parent = 1
0525          do while ( parent .ge. 0 )          ! Find definitions record.
0526          read (unit=data_unit1,err=50000)
0527          1          parent, experiment, ancestor, exps, estim, p
0528          end do
0529          c
0530          c          Collect experiment definition from ancestor's index record.
0531          c
0532          22000 continue
0533          read (unit=index_unit,keyeq=ancestor,keyid=0,iostat=status)
0534          1          mi_pkey, mi_stat, mi_park, mi_ierr, mi_eerr, mi_serr,
0535          2          mi_dtim, mi_ants, mi_waves, mi_noises, mi_smp, mi_sint,
0536          3          mi_begp, mi_endp, mi_delp, mi_peps, mi_norm, mi_wlen,
0537          4          mi_tol, mi_which, mi_ofile, mi_anosies, mi_irtslim,

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0538      5          mi_zeigs, mi_doas, mi_estype, mi_parp
0539 unlock (unit=index_unit)
0540 if ( status .eq. FOR$IOS_SPERECLOC ) then ! Record locked.
0541     type *, 'Record #', ancestor, ' locked. Waiting 5 min.'
0542     call lib$wait( 300.0 ) ! Wait 5 minutes.
0543     go to 22000
0544 else if ( status .ne. 0 ) then ! Error reading index.
0545     close (unit=index_unit)
0546     close (unit=data_unit1)
0547     type *
0548     type *, 'Error reading index file, experiment #', ancestor, '.'
0549     go to 60620
0550 end if
0551 c
0552 c          Ancestor experiment definition read into wad of variables
0553 c listed above. Open virtual arrays for DOA estimates and corresponding
0554 c experiment numbers, and get DOA estimate and experiment number arrays.
0555 c
0556 waves_lin(1) = mi_waves
0557 c
0558 call open_va( doa_add, waves_lin, 1, dr_type, vstatus, msg )
0559 call exit_va( vstatus, msg )
0560 call open_va( expnum_add, waves_lin, 1, di_type, vstatus, msg )
0561 call exit_va( vstatus, msg )
0562 c
0563 if ( .not. ( readarr( %val( doa_add ), dr_type, mi_waves, 1,
0564 1 data_unit1, 4 ) ) ) go to 50000
0565 if ( .not. ( readarr( %val( expnum_add ), di_type, mi_waves, 1,
0566 1 data_unit1, 8 ) ) ) go to 50000
0567 close (unit=data_unit1)
0568 c
0569 c          Now collect antenna array description, wave descriptions,
0570 c observation samples, noise samples, and covariance matrices as
0571 c required. Finally, form estimate of P matrix as selected by estimation
0572 c type.
0573 c
0574 waves_by_9(1) = mi_waves
0575 waves_by_9(2) = 9
0576 waves_sqr(1) = mi_waves
0577 waves_sqr(2) = mi_waves
0578 ants_lin(1) = mi_ants
0579 ants_by_wvs(1) = mi_ants
0580 ants_by_wvs(2) = mi_waves
0581 ants_sqr(1) = mi_ants
0582 ants_sqr(2) = mi_ants
0583 ants_by_2(1) = mi_ants
0584 ants_by_2(2) = 2
0585 wavs_by_smp(1) = mi_waves
0586 wavs_by_smp(2) = mi_smp
0587 smp_by_ants(1) = mi_smp
0588 smp_by_ants(2) = mi_ants
0589 c
0590 call open_va( wave_add, waves_by_9, 2, dr_type, vstatus, msg )
0591 call exit_va( vstatus, msg )
0592 call open_va( ant_add, ants_by_2, 2, dr_type, vstatus, msg )
0593 call exit_va( vstatus, msg )
0594 call open_va( amat_add, ants_by_wvs, 2, dc_type, vstatus, msg )

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0595      call exit_va( vstatus, msg )
0596      call open_va( obsmat_add, smp_by_ants, 2, dc_type, vstatus, msg )
0597      call exit_va( vstatus, msg )
0598      call open_va( noisemat_add, smp_by_ants, 2, dc_type, vstatus, msg )
0599      call exit_va( vstatus, msg )
0600      call open_va( pvm_add, wavs_by_smp, 2, dc_type, vstatus, msg )
0601      call exit_va( vstatus, msg )
0602      call open_va( resm_add, smp_by_ants, 2, dc_type, vstatus, msg )
0603      call exit_va( vstatus, msg )
0604      call open_va( pl2_add, waves_sqr, 2, dc_type, vstatus, msg )
0605      call exit_va( vstatus, msg )
0606      call open_va( pip_add, waves_sqr, 2, dc_type, vstatus, msg )
0607      call exit_va( vstatus, msg )
0608      call open_va( pmat_add, waves_sqr, 2, dc_type, vstatus, msg )
0609      call exit_va( vstatus, msg )
0610      call open_va( sigvm_add, wavs_by_smp, 2, dc_type, vstatus, msg )
0611      call exit_va( vstatus, msg )
0612      call open_va( robs_add, ants_sqr, 2, dc_type, vstatus, msg )
0613      call exit_va( vstatus, msg )
0614      call open_va( rnoise_add, ants_sqr, 2, dc_type, vstatus, msg )
0615      call exit_va( vstatus, msg )
0616      if ( ( estim.ne.1 ) .and. ( estim.ne.3 ) ) then      ! Spwr. type.
0617          call open_va( yarw_add, ants_lin, 1, dr_type, vstatus, msg )
0618          call exit_va( vstatus, msg )
0619      end if
0620
0621      c
0622      c          Get observation, noise, and signal vector samples from the
0623      c          ancestor experiment's "samples" file.
0624
0625      work_file = 'MUSIC_SAMPL_____DAT'
0626      write (unit=work_file( 12 : 21 ),fmt=90020,err=50200) ancestor
0627      open (unit=data_unit1,file='6000*RT:[CHUCK.PARAMS]//work_file,
0628           1      status='OLD',organization='SEQUENTIAL',access='SEQUENTIAL',
0629           2      form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
0630           3      iostat=status,READONLY,err=50010)
0631      if ( .not. ( readarr( %val( obsmat_add ), dc_type, mi_smp, mi_ants,
0632                          data_unit1, 16 ) ) ) go to 50100
0633      if ( .not. ( readarr( %val( noisemat_add ), dc_type, mi_smp, mi_ants,
0634                          data_unit1, 16 ) ) ) go to 50100
0635      if ( .not. ( readarr( %val( sigvm_add ), dc_type, mi_waves, mi_smp,
0636                          data_unit1, 16 ) ) ) go to 50100
0637      close (unit=data_unit1)
0638
0639      c
0640      c          Get antenna array and wave definitions from the ancestor
0641      c          experiment's "input" file.
0642
0643      work_file( 7 : 11 ) = 'INPUT'
0644      open (unit=data_unit1,file='6000*RT:[CHUCK.PARAMS]//work_file,
0645           1      status='OLD',organization='SEQUENTIAL',access='SEQUENTIAL',
0646           2      form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
0647           3      iostat=status,READONLY,err=50010)
0648      read (unit=data_unit1,err=50100)
0649      if ( .not. ( readarr( %val( ant_add ), dr_type, mi_ants, 2,
0650                          data_unit1, 32 ) ) ) go to 50100
0651      if ( .not. ( readarr( %val( wave_add ), dr_type, mi_waves, 9,
0652                          data_unit1, 32 ) ) ) go to 50100
0653      close (unit=data_unit1)

```


MUSIC_PEST

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```

0652 c
0653 c      Get observation and noise covariance matrices from the ancestor
0654 c      experiment's "eigenvalues" file.
0655 c
0656 c      work_file( 7 : 11 ) = 'EIGVS'
0657 c      open (unit=data_unit1,file='6000*RT:[CHUCK.PARAMS]//work_file,
0658 c           1      status='OLD',organization='SEQUENTIAL',access='SEQUENTIAL',
0659 c           2      form='UNFORMATTED',recordtype='FIXED',recl=64,dispose='KEEP',
0660 c           3      iostat=status,READONLY,err=50010)
0661 c      if ( .not. ( readarr( %val( robs_add ), dc_type, mi_ants, mi_ants,
0662 c                    1      data_unit1, 16 ) ) ) go to 50100
0663 c      if ( .not. ( readarr( %val( rnoise_add ), dc_type, mi_ants, mi_ants,
0664 c                    1      data_unit1, 16 ) ) ) go to 50100
0665 c      close (unit=data_unit1)
0666 c
0667 c      For compensated estimation types, subtract the noise samples
0668 c      from the observation samples (replacing the observation samples with
0669 c      the differences).
0670 c
0671 c      if ( ( estim .eq. 3 )
0672 c           1      .or. ( estim .eq. 5 )
0673 c           2      .or. ( estim .eq. 6 ) ) then
0674 c           call c16_sel_vdiff( %val( obsmat_add ), 1,
0675 c                               1      %val( noisemat_add ), 1, mi_smp * mi_ants,
0676 c                               2      %val( obsmat_add ) )
0677 c      end if
0678 c
0679 c      Form estimate of A matrix.
0680 c
0681 c      call sigmat_estm( %val( ant_add ), mi_ants, %val( doa_add ), mi_waves,
0682 c                    1      mi_wlen, %val( amat_add ) )
0683 c
0684 c      Compute actual sample P matrix from original signal vectors,
0685 c      and L2 P matrix estimate.
0686 c
0687 c      tstatus = lib$init_timer()
0688 c      if ( .not. tstatus ) call lib$stop( %val( tstatus ) )
0689 c
0690 c      call dc_matmpyherm( %val( sigvm_add ), mi_waves, mi_smp,
0691 c                    1      %val( sigvm_add ), mi_waves, %val( pmat_add ) )
0692 c      call sqdc_mat_intdiv( %val( pmat_add ), mi_waves, mi_smp )
0693 c      call music_l2p( %val( amat_add ), mi_ants, mi_waves,
0694 c                    1      %val( robs_add ), %val( rnoise_add ),
0695 c                    2      mi_tol, %val( pl2_add ), l2status )
0696 c
0697 c      tstatus = lib$stat_timer( 2, l2tics )
0698 c      if ( .not. tstatus ) call lib$stop( %val( tstatus ) )
0699 c
0700 c      Form estimates of signal and residual vectors and P matrix as
0701 c      selected by the estimation type specified for the experiment.
0702 c
0703 c      tstatus = lib$init_timer()
0704 c      if ( .not. tstatus ) call lib$stop( %val( tstatus ) )
0705 c
0706 c      if ( l2status .eq. 0 ) then      ! Go on only if L2 estimate succeeded.
0707 c          if ( ( estim .eq. 1 )      ! Enumeration.
0708 c              1      .or. ( estim .eq. 3 ) ) then      ! Compensated enum.

```

MUSIC_PEST

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0709      call music_lpp_each( %val( obsmat_add ), mi_smp, mi_ants,
0710      1      %val( amat_add ), mi_waves, p, mi_peps,
0711      2      mi_norm, mi_tol, %val( pvm_add ),
0712      3      %val( resm_add ), mi_irlslim, mnitlim,
0713      4      atitlim, min_del )
0714      else if ( ( estim .eq. 2 ) ) then ! Sum of powers.
0715      1      .or. ( estim .eq. 5 ) ) then ! Compensated sum pwr.
0716      call music_lpw_yar( %val( obsmat_add ), mi_smp, mi_ants,
0717      1      %val( amat_add ), mi_waves, p, mi_peps,
0718      2      mi_norm, mi_tol, %val( yarw_add ),
0719      3      %val( pvm_add ), %val( resm_add ),
0720      4      mi_irlslim, mnitlim, min_del, 1 )
0721      else if ( ( estim .eq. 4 ) ) then ! Sum of powers loss.
0722      1      .or. ( estim .eq. 6 ) ) then ! Comp'd sum pwr loss.
0723      call music_lpw_yar( %val( obsmat_add ), mi_smp, mi_ants,
0724      1      %val( amat_add ), mi_waves, p, mi_peps,
0725      2      mi_norm, mi_tol, %val( yarw_add ),
0726      3      %val( pvm_add ), %val( resm_add ),
0727      4      mi_irlslim, mnitlim, min_del, 2 )
0728      else
0729      go to 50300
0730      end if
0731      if ( ( estim .eq. 1 ) .or. ( estim .eq. 3 ) ) then ! Enum. type.
0732      call dc_matmpyherm( %val( pvm_add ), mi_waves, mi_smp,
0733      1      %val( amat_add ), mi_waves, %val( plp_add ) )
0734      call sqdc_mat_intdiv( %val( plp_add ), mi_waves, mi_smp )
0735      else
0736      ! Sum of powers type.
0737      call music_lpp_spels( %val( amat_add ), mi_ants, mi_waves,
0738      1      %val( yarw_add ), %val( robs_add ),
0739      2      %val( rnoise_add ), mi_tol,
0740      3      %val( plp_add ), status )
0741      if ( ( status .ne. 0 ) .and. ( mnitlim .gt. 0 ) )
0742      1      mnitlim = - mnitlim - 1
0743      end if
0744      end if
0745      c
0746      tstatus = lib$stat_timer( 2, lptics )
0747      if ( .not. tstatus ) call lib$stop( %val( tstatus ) )
0748      c
0749      Finally report results (even if L2 estimate failed).
0750      c
0751      open (unit=data_unit1,file=found_file,status='OLD',
0752      1      organization='SEQUENTIAL',access='DIRECT',form='UNFORMATTED',
0753      2      recordtype='FIXED',recl=8,dispose='KEEP',READONLY,
0754      3      iostat=status)
0755      call music_p_rep( l2status, status+1, %val( doa_add ),
0756      1      %val( expnum_add ), data_unit1, data_unit2,
0757      2      mi_ofile, mi_dtim, experiment, ancestor, exps,
0758      3      %val( pmat_add ), %val( pl2_add ), mi_waves, p,
0759      4      mi_irlslim, mnitlim, atitlim, %val( plp_add ),
0760      5      l2tics / 100.0d0, lptics / 100.0d0, estim, min_del )
0761      close (unit=data_unit1)
0762      call dump_val( vstatus, msg )
0763      call exit_val( vstatus, msg )
0764      c
0765      Succeeded processing input file. Clear error indicator,
delete used input file, and continue search for more input files.

```

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```

0766 c
0767 50130 continue ! Failed reading data file.
0768 fails = 0
0769 call lib$delete_file( found_file )
0770 go to 21000
0771 c
0772 c***** Error branches. *****
0773 c
0774 c Error opening data file. It may be in use by another program.
0775 c
0776 50000 continue ! Data unit 1 & VA open.
0777 close (unit=data_unit1)
0778 c
0779 50010 continue ! VA open.
0780 call dump_va( vstatus, msg )
0781 call exit_va( vstatus, msg )
0782 c
0783 50020 continue ! Nothing open.
0784 if ( fails .ge. fails_lim ) then ! Failed too many times.
0785 type *,
0786 type *, 'Failed opening input data file(s) ', fails_lim, ' times.'
0787 go to 60620
0788 else
0789 fails = fails + 1 ! Haven't failed too many.
0790 go to 21000
0791 end if
0792 c
0793 c Error reading opened intermediate data file.
0794 c
0795 50100 continue
0796 close (unit=data_unit1)
0797 call dump_va( vstatus, msg )
0798 call exit_va( vstatus, msg )
0799 c
0800 type *
0801 type *, 'Failure reading data file: ', work_file
0802 50110 continue
0803 type *, 'while processing input file:'
0804 type *, found_file
0805 type *
0806 type *, 'Deleting the input file.'
0807 go to 50130
0808 c
0809 c Failure performing internal write.
0810 c
0811 50200 continue
0812 type *
0813 type *, 'Failed performing internal write.'
0814 go to 60620
0815 c
0816 c Invalid estimation method.
0817 c
0818 50300 continue
0819 call dump_va( vstatus, msg )
0820 call exit_va( vstatus, msg )
0821 c
0822 type *

```

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0823          type *, 'Invalid estimation method encountered'
0824          go to 50110
0825          c
0826          90020 format( i10.10 )
0827          c
0828          end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	3041	PIC CON REL LCL SHR EXE RD NOWRT L
1 \$PDATA	616	PIC CON REL LCL SHR NOEXE RD NOWRT L
2 \$LOCAL	2660	PIC CON REL LCL NOSHR NOEXE RD WRT C
3 MUSIC_MEASURES	120	PIC OVR REL GBL SHR NOEXE RD WRT L
Total Space Allocated		6437

ENTRY POINTS

Address	Type	Name
0-00000000		MUSIC_PEST

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type	Name
2-000002D4	I*4	AMAT_ADD	2-000002B8	I*4	ANCESTOR	**	I*4	ANT:
2-00000308	I*4	ATITLIM	3-0000000C	I*4	CYCLES	2-00000290	I*4	DAT:
2-000002C4	I*4	DOA_ADD	2-000002C0	I*4	ESTIM	2-000002B4	I*4	EXP:
2-0000028C	I*4	EXPS	**	I*4	FAILS	**	I*4	FAIL
2-00000288	L*4	FOUND	2-0000010D	CHAR	FOUND_FILE	2-0000028C	I*4	INDI
2-0000030C	I*4	L2TICS	2-00000310	I*4	LPTICS	2-00000240	R*8	MIN:
2-00000250	I*4	MI_ANTS	2-00000278	R*4	MI_BEGP	2-00000280	R*4	MI_I
2-00000041	CHAR	MI_DTIM	2-0000020E	I*2	MI_EERR	2-0000027C	R*4	MI_I
2-0000020C	I*2	MI_IERR	2-00000268	I*4	MI_IRLSLIM	2-00000258	I*4	MI_I
2-00000054	CHAR	MI_OFILE	2-0000024C	I*4	MI_PARK	2-00000284	R*4	MI_I
2-00000248	I*4	MI_PKEY	2-00000210	I*2	MI_SERR	2-00000218	R*8	MI_I
2-00000040	CHAR	MI_STAT	2-00000238	R*8	MI_TOL	2-00000254	I*4	MI_I
2-00000230	R*8	MI_WLEN	2-0000026C	I*4	MI_ZEIGS	2-00000304	I*4	MNI:
2-000002DC	I*4	NOISEMAT_ADD	3-00000060	R*8	NON_TRI	3-00000068	I*4	NON
2-00000314	R*4	P	2-00000280	I*4	PARENT	2-000002E8	I*4	PL2:
2-000002F0	I*4	PMAT_ADD	2-000002E0	I*4	PVM_ADD	3-0000006C	R*8	QPS
2-000002E4	I*4	RESM_ADD	2-000002FC	I*4	RNOISE_ADD	2-000002F8	I*4	ROB:
2-000002F4	I*4	SIGVM_ADD	2-00000298	I*4	SPAWN_UNIT	**	CHAR	SP_I
**	CHAR	SP_PROC	**	I*4	SP_STAT	2-0000029C	I*4	STA
3-00000000	R*8	TOL	**	I*4	TSTATUS	3-00000018	R*8	UPH
3-00000020	R*8	UPHI_SMARG	3-00000034	I*4	UPHI_S_ZCYC	3-00000028	R*8	U_U:
3-0000003C	R*8	VPSI_LMARG	3-00000054	I*4	VPSI_L_ZCYC	3-00000044	R*8	VPS
2-000002A0	I*4	VSTATUS	3-0000004C	R*8	V_UNIERR	3-0000005C	I*4	V_U:
**	I*4	WAITS_LIM	**	I*4	WAVES	2-000002CC	I*4	WAV:

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14-Apr-1988 08:40:56 BRA

```

0001      subroutine bracket_doa( doa, apars, num_ants, wavelen, eigvec,
0002      1                          num_zeigs, p, down, pwr, lower, upper, perf )
0003      c
0004      c-----
0005      c
0006      c          This routine computes the sharpness measure for a peak in a
0007      c          MUSIC "power" spectrum.
0008      c
0009      c          Parameters:
0010      c          -----
0011      c          num_ants:  INTEGER*4 number of antenna elements.
0012      c          num_zeigs: INTEGER*4 number of eigenvectors to use in
0013      c                      computing the MUSIC power spectrum.
0014      c          p:         REAL*4 value of the exponent to which to raise the terms in
0015      c                      the MUSIC power spectrum expression
0016      c                      ( 1 <= p <= 3 ).
0017      c          doa:      REAL*8 direction of arrival estimate (RADIANS) of the
0018      c                      peak for which the sharpness measure is
0019      c                      to be computed. This need not actually
0020      c                      identify a peak location; the routine
0021      c                      locates the lower and upper angles
0022      c                      which produce MUSIC spectrum values
0023      c                      that are (down)db down from the value
0024      c                      at angle doa.
0025      c          apars:    num_ants row REAL*8 array of antenna element
0026      c                      parameters. The first column of apars
0027      c                      contains element ranges from the origin
0028      c                      and the second column contains element
0029      c                      angles (in RADIANS) from the abscissa.
0030      c          wavelen:  REAL*8 carrier wavelength to be used in computing the
0031      c                      MUSIC power spectrum. This must be in
0032      c                      the same units as the antenna distances
0033      c                      from the origin (apars(*,1)).
0034      c          down:     REAL*8 number of db down from the MUSIC spectrum peak
0035      c                      value at which the lower and upper
0036      c                      "bracket" angles are to be identified.
0037      c          eigvec:   num_ants x num_ants COMPLEX*16 matrix whose first
0038      c                      num_zeigs columns are the "noise"
0039      c                      eigenvectors to be used for computing
0040      c                      MUSIC power spectrum values.
0041      c          pwr:      REAL*8 output value of the MUSIC spectrum at angle doa.
0042      c          lower:    REAL*8 output value of the largest angle (RADIANS) less
0043      c                      than doa (but >= - pi) at which the
0044      c                      MUSIC spectrum value is (down)db down
0045      c                      from pwr.
0046      c          upper:    REAL*8 output value of the smallest angle (RADIANS)
0047      c                      greater than doa (but <= pi) at which
0048      c                      the MUSIC spectrum value is (down)db
0049      c                      down from pwr.
0050      c          perf:     REAL*8 output value of the sharpness measure for the
0051      c                      peak at doa.
0052      c
0053      c-----
0054      c
0055      c          implicit none
0056      c          intrinsic dlog10
0057      c          external open_va, exit_va, free_va

```

BRACKET_DOA

18-Apr-1988 14:12:16

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BRACKI

```

0058      external music_power, sigval
0059      include '6000*RT:[CHUCK.RESEARCH.FORTDIR]VATYPES.TXT'
0071      include '(real_constants)'
0122      c
0123      real*8 doadel
0124      parameter ( doadel = rad_p_deg8 )
0125      c
0126      real*8 music_power
0127      c
0128      integer*4 num_ants, num_zeigs
0129      real*4 p
0130      real*8 doa, apars( num_ants, 1 ), pwr, lower
0131      real*8 wavelen, upper, perf, down
0132      complex*16 eigvec( num_ants, num_ants )
0133      c
0134      integer*4 steerv_add, linear(1), status
0135      real*8 dpwr
0136      character*80 msg
0137      c
0138      linear(1) = num_ants
0139      call open_va( steerv_add, linear, 1, dc_type, status, msg )
0140      call exit_va( status, msg )
0141      c
0142      call sigval( apars, num_ants, doa, wavelen, %val( steerv_add ) )
0143      pwr = music_power( eigvec, %val( steerv_add ), num_ants, num_zeigs, p )
0144      dpwr = pwr * ( 10.0d0 ** ( - down / 10.0d0 ) )
0145      c
0146      lower = doa - doadel
0147      do while ( lower .gt. - pi8 )
0148          call sigval( apars, num_ants, lower, wavelen, %val( steerv_add ) )
0149          if ( music_power( eigvec, %val( steerv_add ), num_ants,
0150              1                                     num_zeigs, p )
0151              .le. dpwr ) go to 10000
0152          lower = lower - doadel
0153      end do
0154      10000 continue
0155      c
0156      upper = doa + doadel
0157      do while ( upper .lt. pi8 )
0158          call sigval( apars, num_ants, upper, wavelen, %val( steerv_add ) )
0159          if ( music_power( eigvec, %val( steerv_add ), num_ants,
0160              1                                     num_zeigs, p )
0161              .le. dpwr ) go to 10010
0162          upper = upper + doadel
0163      end do
0164      10010 continue
0165      c
0166      if ( upper .gt. lower ) then
0167          perf = dlog10( pwr ) / ( p * ( upper - lower ) )
0168      else
0169          perf = -1.0d0
0170      end if
0171      c
0172      call free_va( steerv_add, status, msg )
0173      call exit_va( status, msg )
0174      c
0175      return

```

18-Apr-1988 14:12:53 U
28-Mar-1988 20:34:40 C

```

0001      subroutine c16_vachol( inmat, size, outmat, condit, status )
0002      c
0003      c -----
0004      c
0005      c      This subroutine uses LINPAC routine ZPOCO, and routines from
0006      c      the Virtual Array system (copyright 1987 by Dwight Day), to perform a
0007      c      Cholesky decomposition of the COMPLEX*16 size x size input matrix
0008      c      inmat. The Cholesky factor is returned in the size x size COMPLEX*16
0009      c      output matrix outmat. If the factorization succeeded (inmat is
0010      c      positive definite), the INTEGER*4 parameter status is set to 0 and the
0011      c      REAL*8 parameter condit is set to an estimate of the inverse of the
0012      c      condition number of inmat. If factorization failed, status is set to
0013      c      a value > 0 (size of the leading submatrix of inmat that is not
0014      c      positive definite), and condit is not changed.
0015      c      The Cholesky decomposition of inmat produces an upper
0016      c      triangular matrix outmat such that:
0017      c      inmat = outmat-conjugate-transpose * outmat.
0018      c      Since inmat is Hermitian, the strictly lower triangle of inmat
0019      c      is NEVER REFERENCED. The strictly lower triangle of outmat is SET TO
0020      c      ZERO. The input matrix inmat is NOT CHANGED by this routine.
0021      c -----
0022      c
0023      c
0024      c      implicit none
0025      c      external zpoco, open_va, exit_va, free_va
0026      c      include '6000*RT:[CHUCK.RESEARCH.FORTDIR]VATYPES.TXT'
0027      c
0028      c
0029      c      integer*4 size, status
0030      c      real*8 condit
0031      c      complex*16 inmat( size, size ), outmat( size, size )
0032      c
0033      c      integer*4 i, j
0034      c      integer*4 work_add, size_lin( 1 ), vstatus
0035      c      character*80 msg
0036      c
0037      c      Copy input upper triangle to output upper triangle, since
0038      c      ZPOCO puts its output in its input.
0039      c
0040      c
0041      c      do i = 1, size
0042      c      do j = 1, i - 1
0043      c      outmat(i,j) = ( 0.0d0, 0.0d0 )
0044      c      end do
0045      c      do j = i, size
0046      c      outmat(i,j) = inmat(i,j)
0047      c      end do
0048      c      end do
0049      c
0050      c      Make a work vector for ZPOCO.
0051      c
0052      c      size_lin(1) = size
0053      c      call open_va( work_add, size_lin, 1, dc_type, vstatus, msg )
0054      c      call exit_va( vstatus, msg )
0055      c
0056      c      Now let ZPOCO do the real work.
0057      c
0058      c      call zpoco( outmat, size, size, condit, %val( work_add ), status )
0059      c
0060      c
0061      c
0062      c
0063      c
0064      c
0065      c
0066      c
0067      c
0068      c

```

18-Apr-1988 14:12:53 U
23-Mar-1988 20:34:40 C

```

0001      subroutine c16_vachol( inmat, size, outmat, condit, status )
0002      c
0003      c -----
0004      c
0005      c      This subroutine uses LINPAC routine ZPOCO, and routines from
0006      c      the Virtual Array system (copyright 1987 by Dwight Day), to perform a
0007      c      Cholesky decomposition of the COMPLEX*16 size x size input matrix
0008      c      inmat. The Cholesky factor is returned in the size x size COMPLEX*16
0009      c      output matrix outmat. If the factorization succeeded (inmat is
0010      c      positive definite), the INTEGER*4 parameter status is set to 0 and the
0011      c      REAL*8 parameter condit is set to an estimate of the inverse of the
0012      c      condition number of inmat. If factorization failed, status is set to
0013      c      a value > 0 (size of the leading submatrix of inmat that is not
0014      c      positive definite), and condit is not changed.
0015      c      The Cholesky decomposition of inmat produces an upper
0016      c      triangular matrix outmat such that:
0017      c      inmat = outmat-conjugate-transpose * outmat.
0018      c      Since inmat is Hermitian, the strictly lower triangle of inmat
0019      c      is NEVER REFERENCED. The strictly lower triangle of outmat is SET TO
0020      c      ZERO. The input matrix inmat is NOT CHANGED by this routine.
0021      c -----
0022      c
0023      c
0024      c      implicit none
0025      c      external zpoco, open_va, exit_va, free_va
0026      c      include '6000*RT:[CHUCK.RESEARCH.FORTDIR]VATYPES.TXT'
0027      c
0028      c
0029      c      integer*4 size, status
0030      c      real*8 condit
0031      c      complex*16 inmat( size, size ), outmat( size, size )
0032      c
0033      c      integer*4 i, j
0034      c      integer*4 work_add, size_lin( 1 ), vstatus
0035      c      character*80 msg
0036      c
0037      c      Copy input upper triangle to output upper triangle, since
0038      c      ZPOCO puts its output in its input.
0039      c
0040      c
0041      c      do i = 1, size
0042      c      do j = 1, i - 1
0043      c      outmat(i,j) = ( 0.0d0, 0.0d0 )
0044      c      end do
0045      c      do j = i, size
0046      c      outmat(i,j) = inmat(i,j)
0047      c      end do
0048      c      end do
0049      c
0050      c      Make a work vector for ZPOCO.
0051      c
0052      c      size_lin(1) = size
0053      c      call open_va( work_add, size_lin, 1, dc_type, vstatus, msg )
0054      c      call exit_va( vstatus, msg )
0055      c
0056      c      Now let ZPOCO do the real work.
0057      c
0058      c      call zpoco( outmat, size, size, condit, %val( work_add ), status )
0059      c
0060      c
0061      c
0062      c
0063      c
0064      c
0065      c
0066      c
0067      c
0068      c

```


18-Apr-1988 14:13:22 VAX
24-Sep-1987 11:45:08 C8_

```

0001      subroutine c8_noise( noise, size, defs, pars )
0002      c
0003      c-----
0004      c
0005      c      This subroutine computes a COMPLEX*8 noise "sources" vector.
0006      c
0007      c      Parameters:
0008      c      -----
0009      c      noise = COMPLEX*8 size-element output noise vector.
0010      c      size = INTEGER*4 number of noise sources in use.
0011      c      defs = INTEGER*4 size x 2 matrix whose first column is the
0012      c      vector of noise source types (see subroutine
0013      c      READ_NOISE), and whose second column is the
0014      c      vector of counts for intermittent noise
0015      c      sources.
0016      c      pars = REAL*4 size x 3 matrix of noise source parameters (see
0017      c      subroutine READ_NOISE).
0018      c
0019      c-----
0020      c
0021      c      implicit none
0022      c      intrinsic cmplx, jshft, jland
0023      c      external gauss, uni
0024      c
0025      c      real*4 uni
0026      c
0027      c      integer*4 size, defs( size, 1 )
0028      c      real*4 pars( size, 1 )
0029      c      complex*8 noise( size )
0030      c
0031      c      integer*4 i, it1
0032      c      real*4 rnoise, inoise
0033      c
0034      c      do i = 1, size
0035      c          it1 = jshft( defs(i,1), -16 )
0036      c          if ( ( ( it1 .ne. 0 ) .and. ( defs(i,2) .eq. 1 ) ) .or.
0037      c              ( it1 .eq. 0 ) ) then
0038      c              it1 = jland( defs(i,1), 65536 )
0039      c              if ( it1 .eq. 1 ) then
0040      c                  noise(i) = cmplx( pars(i,1), pars(i,2) )
0041      c              else if ( it1 .eq. 2 ) then
0042      c                  noise(i) = cmplx( pars(i,1) + pars(i,2) * ( uni( 10571 ) - 0.5 ),
0043      c                      1      pars(i,1) + pars(i,2) * ( uni( 10571 ) - 0.5 )
0044      c                      2      )
0045      c              else
0046      c                  call gauss( rnoise, inoise, pars(i,2), pars(i,1), 0 )
0047      c                  noise(i) = cmplx( rnoise, inoise )
0048      c              end if
0049      c          else
0050      c              noise(i) = ( 0.0, 0.0 )
0051      c          end if
0052      c      end do
0053      c
0054      c      return
0055      c      end

```

18-Apr-1988 14:13:58 Vr
16-Mar-1988 09:30:14 Dc

```

0001      subroutine dcw_irls( a, n, m, x, p, eps, norm,
0002      1                                tol, itlim, f, resid, del )
0003      c
0004      c -----
0005      c
0006      c      This subroutine forms an approximate Lp estimate of a vector
0007      c      f which satisfies the equation  $x = a f + \text{noise}$ . The IRLS algorithm
0008      c      is used to determine the f estimate.
0009      c
0010      c      Parameters:
0011      c      -----
0012      c      a:  n x m COMPLEX*16 coefficient matrix.
0013      c      n:  INTEGER*4 number of rows in the coefficient matrix (usually
0014      c          the number of observations obtained).
0015      c      m:  INTEGER*4 number of elements in the vector to be estimated
0016      c          (problem order).
0017      c      x:  n-element COMPLEX*16 observation vector.
0018      c      p:  REAL*4 power of error magnitudes to be minimized (an Lp
0019      c          estimate is computed).
0020      c      eps: REAL*8 minimum residual magnitude to be used (see
0021      c          subroutine get_w).
0022      c      norm: REAL*8 value to which to normalize the SQUARE ROOT OF
0023      c          the maximum weight. If norm <= 0.0d0,
0024      c          no normalization occurs. Otherwise,
0025      c          all weight SQUARE ROOT values (w(i))
0026      c          are divided by a value which results in
0027      c          a maximum weight SQUARE ROOT of norm.
0028      c      tol: REAL*8 value to be taken for zero. The routine exits
0029      c          when the SQUARE of the norm of the
0030      c          difference between successive f
0031      c          approximations drops below tol. NOTE
0032      c          that tol expresses a FRACTIONAL
0033      c          tolerance; that is, || difference || /
0034      c          || sum || <= tol is the exit criterion.
0035      c      itlim: INTEGER*4 number of iterations permitted. This
0036      c          variable is COUNTED DOWN at each
0037      c          iteration, and the routine exits if it
0038      c          reaches zero (regardless of
0039      c          convergence). Also, the variable is
0040      c          SET to the negative of its current
0041      c          value minus one and the routine exits
0042      c          IMMEDIATELY if the routine DC_MPPSINV
0043      c          fails ("info", below, returned with a
0044      c          non-zero value).
0045      c      f:  m-element COMPLEX*16 computed estimate vector.
0046      c      resid: n-element vector of residuals from the last computed
0047      c          estimate. resid = x - a f.
0048      c      del: REAL*8 exit value of the square of the norm of the
0049      c          difference between successive estimates
0050      c          of f.
0051      c
0052      c      NOTE that an L2 estimate of f is produced if this routine is
0053      c      called with itlim = 0 (regardless of the passed value of p).
0054      c
0055      c -----
0056      c
0057      c      implicit none

```

```

DCV_IRLS                                     18-Apr-1988 14:13:58      VAX FORT
                                              16-Mar-1988 09:30:14      DCV_IRLS

0058      external open_va, free_va, exit_va
0059      external dc_mppsinv, dc_matvmpy, residuals, dc_veccopy, get_w_half
0060      external drdi_dc_mmpy, drdi_dc_vmpy, dc_vdel_nmsq
0061      c
0062      real*8 dc_vdel_nmsq
0063      c
0064      integer*4 n, m, itlim
0065      real*4 p
0066      real*8 eps, norm, tol, del
0067      complex*16 a( n, m ), x( n ), f( m ), resid( n )
0068      c
0069      integer*4 prevf_add, mlin( 1 ), dc_type/7/, status
0070      integer*4 w_add, nlin( 1 ), dr_type/5/
0071      integer*4 psinv_add, m_by_n( 2 )
0072      integer*4 wx_add, wa_add, n_by_m( 2 )
0073      integer*4 info
0074      character*80 msg
0075      c
0076      mlin(1) = m
0077      nlin(1) = n
0078      m_by_n(1) = m
0079      m_by_n(2) = n
0080      n_by_m(1) = n
0081      n_by_m(2) = m
0082      c
0083      c          Get space for various working arrays.
0084      c
0085      call open_va( prevf_add, mlin, 1, dc_type, status, msg )
0086      call exit_va( status, msg )
0087      call open_va( w_add, nlin, 1, dr_type, status, msg )
0088      call exit_va( status, msg )
0089      call open_va( psinv_add, m_by_n, 2, dc_type, status, msg )
0090      call exit_va( status, msg )
0091      call open_va( wx_add, nlin, 1, dc_type, status, msg )
0092      call exit_va( status, msg )
0093      call open_va( wa_add, n_by_m, 2, dc_type, status, msg )
0094      call exit_va( status, msg )
0095      c
0096      c          Compute L2 estimate of f, and residuals therefor.
0097      c
0098      del = tol + .5
0099      call dc_mppsinv( a, n, m, tol, %val( psinv_add ), info )
0100      if ( info .ne. 0 ) then                                ! Error in psuedo-inverse.
0101          itlim = - itlim - 1
0102          go to 10000
0103      end if
0104      call dc_matvmpy( %val( psinv_add ), m, n, x, f )
0105      call residuals( x, n, a, m, f, resid )
0106      c
0107      c          Exit if L2 solution desired (itlim = 0), otherwise start
0108      c          iterating to Lp solution.
0109      c
0110      do while ( ( del .gt. tol ) .and. ( itlim .gt. 0 ) )
0111          call dc_veccopy( f, m, %val( prevf_add ) )
0112      c
0113          call get_w_half( resid, n, p, eps, norm, %val( w_add ) )
0114          ! Note using SQRT( weights ).

```

DCV_IRLS

18-Apr-1988 14:13:58
16-Mar-1988 09:30:14VAX F
DCV_I

```

0115      call drdi_dc_mmpy( %val( w_add ), n, a, m, %val( wa_add ) )
0116      call dc_mppsinv( %val( wa_add ), n, m, tol, %val( psinv_add ), info )
0117      if ( info .ne. 0 ) then
0118          itlim = - itlim - 1
0119          go to 10000
0120      end if
0121      call drdi_dc_vmpy( %val( w_add ), n, x, %val( wx_add ) )
0122      call dc_matvmpy( %val( psinv_add ), m, n, %val( wx_add ), f )
0123      call residuals( x, n, a, m, f, resid )
0124  c
0125      itlim = itlim - 1
0126      del = dc_vdel_nmsq( f, %val( prevf_add ), m, tol )
0127  end do
0128  c
0129      Dump the working array space.
0130  c
0131  10000 continue
0132      call free_va( prevf_add, status, msg )
0133      call exit_va( status, msg )
0134      call free_va( w_add, status, msg )
0135      call exit_va( status, msg )
0136      call free_va( psinv_add, status, msg )
0137      call exit_va( status, msg )
0138      call free_va( wx_add, status, msg )
0139      call exit_va( status, msg )
0140      call free_va( wa_add, status, msg )
0141      call exit_va( status, msg )
0142  c
0143      return
0144  end

```

18-Apr-1988 14:13:58 VA)
16-Mar-1988 09:30:14 DC\

```
0001 c
0002 c      subroutine residuals( x, n, a, m, f, resid )
0003 c
0004 c-----
0005 c
0006 c      This subroutine calculates the n-element COMPLEX*16 vector
0007 c      (resid) of residuals for an estimate of f. x is n-element COMPLEX*16,
0008 c      a is n x m COMPLEX*16, n and m are INTEGER*4, and f is m-element
0009 c      COMPLEX*16.
0010 c      The formula used is resid = x - a f.
0011 c
0012 c-----
0013 c
0014 c      implicit none
0015 c      external dc_matvmpy
0016 c
0017 c      integer*4 n, m
0018 c      complex*16 x( n ), a( n, m ), f( m ), resid( n )
0019 c
0020 c      integer*4 i
0021 c
0022 c      call dc_matvmpy( a, n, m, f, resid )
0023 c      do i = 1, n
0024 c          resid(i) = x(i) - resid(i)
0025 c      end do
0026 c
0027 c      return
0028 c      end
```

18-Apr-1988 14:15:16 VA)
17-Nov-1987 08:45:17 DC.

```

0001      subroutine dc_mppsinv( a, n, m, fuzz, psinv, info )
0002      c
0003      c-----
0004      c
0005      c          This subroutine forms the Moore-Penrose psuedo-inverse of a
0006      c          COMPLEX*16 matrix. A Singular Value Decomposition (SVD) algorithm
0007      c          is used, so that no explicit inversion is required; thus the matrix to
0008      c          be "inverted" need not be of full rank (or, for that matter, square).
0009      c          The Moore-Penrose psuedo-inverse of a is computed as:
0010      c          psinv = ( a_conjugate_transpose * a )-inverse * a_conjugate_transpose.
0011      c
0012      c          Parameters:
0013      c          -----
0014      c          a:      n x m COMPLEX*16 matrix for which the psuedo-inverse is to
0015      c                  be computed.
0016      c          n, m:   INTEGER*4 number of rows and columns, respectively, in
0017      c                  the a matrix.
0018      c          fuzz:  REAL*8 value to be taken as zero ("tolerance") (used
0019      c                  to determine when a singular value is
0020      c                  "almost" zero).
0021      c          psinv: m x n COMPLEX*16 computed psuedo-inverse.
0022      c          info:  INTEGER*4 status indicator (output); 0 for successful
0023      c                  psuedo-inversion, non-0 on failure of
0024      c                  singular value decomposition.
0025      c
0026      c-----
0027      c
0028      c          implicit none
0029      c          intrinsic jmin0
0030      c          external open_va, free_va, exit_va
0031      c          external zsvdc
0032      c          external dc_matcopy, dc_matmpyherm, dcsv_vsigm1
0033      c
0034      c          integer*4 n, m, info
0035      c          real*8 fuzz
0036      c          complex*16 a( n, m ), psinv( m, n )
0037      c
0038      c          integer*4 dc_type//, status
0039      c          integer*4 n_lin( 1 ), n_sqr( 2 ), m_lin( 1 ), m_sqr( 2 )
0040      c          integer*4 n_by_m( 2 ), m_by_n( 2 ), slen( 1 )
0041      c          integer*4 aa_add, s_add, e_add, u_add, v_add, wk1_add, wk2_add
0042      c          character*80 msg
0043      c
0044      c          n_lin(1) = n
0045      c          n_sqr(1) = n
0046      c          n_sqr(2) = n
0047      c          m_lin(1) = m
0048      c          m_sqr(1) = m
0049      c          m_sqr(2) = m
0050      c          n_by_m(1) = n
0051      c          n_by_m(2) = m
0052      c          m_by_n(1) = m
0053      c          m_by_n(2) = n
0054      c          slen(1) = jmin0( n + 1, m )
0055      c
0056      c          call open_va( aa_add, n_by_m, 2, dc_type, status, msg )
0057      c          call exit_va( status, msg )

```

DC_MPPS INV

18-Apr-1988 14:15:16

17-Nov-1987 08:45:17

```

0058      call open_va( s_add, slen, 1, dc_type, status, msg )
0059      call exit_va( status, msg )
0060      call open_va( e_add, m_lin, 1, dc_type, status, msg )
0061      call exit_va( status, msg )
0062      call open_va( u_add, n_sqr, 2, dc_type, status, msg )
0063      call exit_va( status, msg )
0064      call open_va( v_add, m_sqr, 2, dc_type, status, msg )
0065      call exit_va( status, msg )
0066      call open_va( wk1_add, n_lin, 1, dc_type, status, msg )
0067      call exit_va( status, msg )
0068      call open_va( wk2_add, m_by_n, 2, dc_type, status, msg )
0069      call exit_va( status, msg )
0070      c
0071      call dc_matcopy( a, n, m, %val( aa_add ) )
0072      call zsvdc( %val( aa_add ), n, n, m, %val( s_add ), %val( e_add ),
0073              1 %val( u_add ), n, %val( v_add ), m,
0074              2 %val( wk1_add ), 11, info )
0075      if ( info .ne. 0 ) go to 10000
0076      c
0077      call dcsv_vsigm1( %val( v_add ), m, %val( s_add ), n, fuzz,
0078                    1 %val( wk2_add ) )
0079      call dc_matmpyherm( %val( wk2_add ), m, n, %val( u_add ), n, psinv )
0080      c
0081      10000      continue
0082      call free_va( aa_add, status, msg )
0083      call exit_va( status, msg )
0084      call free_va( s_add, status, msg )
0085      call exit_va( status, msg )
0086      call free_va( e_add, status, msg )
0087      call exit_va( status, msg )
0088      call free_va( u_add, status, msg )
0089      call exit_va( status, msg )
0090      call free_va( v_add, status, msg )
0091      call exit_va( status, msg )
0092      call free_va( wk1_add, status, msg )
0093      call exit_va( status, msg )
0094      call free_va( wk2_add, status, msg )
0095      call exit_va( status, msg )
0096      c
0097      return
0098      end

```


18-Apr-1988 14:17:03
22-Jun-1987 20:01:56

```

0001      subroutine eig_from_sv( svb, n, fuzz, eig )
0002      c
0003      c-----
0004      c
0005      c          This subroutine completes the operations started by subroutine
0006      c      sq_vlgsvd, producing the generalized eigenvalues for the matrix pair
0007      c      (a, b).
0008      c
0009      c      Parameters:
0010      c      -----
0011      c          svb: n-element COMPLEX*16 vector of singular values of
0012      c                  b * a_inv (all entries should be real).
0013      c          n:   INTEGER*4 number of elements in svb and eig.
0014      c          fuzz: REAL*8 value to be taken for zero ("tolerance").
0015      c          eig: n-element REAL*8 result vector of generalized
0016      c                  eigenvalues.
0017      c
0018      c-----
0019      c
0020      c      implicit none
0021      c      intrinsic dreal
0022      c
0023      c      integer*4 n
0024      c      real*8 fuzz, eig(n)
0025      c      complex*16 svb(n)
0026      c
0027      c      integer*4 i
0028      c      real*8 svsqb
0029      c
0030      c          Compute generalized eigenvalues.
0031      c
0032      c      do i = 1, n
0033      c          svsqb = dreal( svb(i) )
0034      c          svsqb = svsqb * svsqb
0035      c          if ( svsqb .le. fuzz ) then
0036      c              eig(i) = 0.0d0
0037      c          else
0038      c              eig(i) = 1.0d0 / svsqb
0039      c          end if
0040      c      end do
0041      c
0042      c      return
0043      c      end

```

18-Apr-1988 14:17:39 VA:
2-Apr-1988 16:47:00 EN:

```

0001      logical function endspawn_if( log_unit, prog_num )
0002      c
0003      c-----
0004      c
0005      c          This function reads (and locks) the record with primary key
0006      c          value prog_num in the indexed-organization file open on logical unit
0007      c          number log_unit.  If the specified record is locked when the read
0008      c          attempt is made, the function exits returning a value of .false..
0009      c          If the desired record has been read and locked, the function
0010      c          checks that the process named in the record is not present on the VAX,
0011      c          and if it is present exits returning the value .false..  If the
0012      c          process is not present, the function sets the "spawned" indication in
0013      c          the record to 0, rewrites (and unlocks) the record, and returns .true..
0014      c          If an error occurs during any of the system calls, the function
0015      c          prints an error message and returns .true..
0016      c          The FORTRAN program MAKE_SPAWN_IDX can be used to create a file
0017      c          of the required format.  The text file
0018      c          6000*RT:[CHUCK.RESEARCH.FORTDIR]SPAUNED_DEF.TXT contains the
0019      c          definitions required to access such a file.  The parameters log_unit
0020      c          and prog_num are INTEGER*4.
0021      c
0022      c-----
0023      c
0024      c          implicit none
0025      c          external lib$getjpi, lib$wait
0026      c          include '(system_symbols)'
0027      c          include 'SYS$LIBRARY:FORIOSDEF'
0028      c          include '6000*RT:[CHUCK.RESEARCH.FORTDIR]SPAUNED_DEF.TXT'
0029      c
0030      c          integer*4 lib$getjpi
0031      c
0032      c          integer*4 log_unit, prog_num
0033      c
0034      c          integer*4 status, out_int
0035      c          character*30 err_type
0036      c
0037      c          read (unit=log_unit,keyeq=prog_num,keyid=0,iostat=status)
0038      c          1          sp_num, sp_stat, sp_file, sp_proc
0039      c          if      ( status .eq. FOR$IOS_SPERECLOC ) then ! Record lock.
0040      c              unlock (unit=log_unit)
0041      c              endspawn_if = .false.
0042      c              return
0043      c          else if ( status .ne. 0 ) then          ! Read error.
0044      c              type *
0045      c              type *, 'Error during read, code: ', status
0046      c              type *
0047      c              unlock (unit=log_unit)
0048      c              endspawn_if = .true.
0049      c              return
0050      c          end if
0051      c
0052      c          status = lib$getjpi( JPI$_PID, , sp_proc, out_int )
0053      c          if      ( ( status .eq. SS$_NORMAL )
0054      c              .or. ( status .eq. SS$_SUSPENDED ) ) then
0055      c              unlock (unit=log_unit)
0056      c              endspawn_if = .false.
0057      c              return

```

ENDSPAWN_IF

18-Apr-1988 14:17:39

2-Apr-1988 16:47:00

```

0302     else if ( status .eq. SS$_NONEXPR ) then
0303         go to 10000
0304     else if ( status .eq. SS$_BADPARAM ) then
0305         unlock (unit=log_unit)
0306         err_type = 'Bad parameter.'
0307     else if ( status .eq. SS$_ACCVIO ) then
0308         unlock (unit=log_unit)
0309         err_type = 'Access violation.'
0310     else if ( status .eq. SS$_IVLOGNAM ) then
0311         unlock (unit=log_unit)
0312         err_type = 'Invalid process name length.'
0313     else if ( status .eq. SS$_NOMOREPROC ) then
0314         unlock (unit=log_unit)
0315         err_type = 'No "more" processes.'
0316     else if ( status .eq. SS$_NOPRIV ) then
0317         unlock (unit=log_unit)
0318         err_type = 'No privilege.'
0319     else
0320         unlock (unit=log_unit)
0321         err_type = 'Code:          hex.'
0322         write (unit=err_type( 8 : 15 ),fmt=90000) status
0323     end if
0324     c
0325     type *
0326     type *, 'Failed checking "', sp_proc, '" : ', err_type
0327     type *
0328     endspawn_if = .true.
0329     return
0330     c
0331     10000 continue
0332     sp_stat = 0
0333     rewrite (unit=log_unit,err=20000,iostat=status)
0334     1          sp_num, sp_stat, sp_file, sp_proc
0335     unlock (unit=log_unit)
0336     c
0337     endspawn_if = .true.
0338     return
0339     c
0340     20000 continue          ! Rewrite error.
0341     unlock (unit=log_unit)
0342     type *
0343     type *, 'Error during rewrite, code: ', status
0344     type *
0345     endspawn_if = .true.
0346     return
0347     c
0348     90000 format( z3.8 )
0349     c
0350     end

```

18-Apr-1988 14:18:06
25-Feb-1988 18:45:56

```

0001      subroutine get_music_def( num_ants, num_waves, noises, anoises,
0002      1                                which, num_smp )
0003      c
0004      c-----
0005      c
0006      c          This subroutine sets the number of antennae, waves, samples,
0007      c          and noise generators to use for the MUSIC / Lp algorithm based on
0008      c          on operator input. The default values are assumed to be the values o
0009      c          entry to the routine.
0010      c
0011      c          Parameters:
0012      c          -----
0013      c          num_ants:  INTEGER*4 number of antenna array elements.
0014      c          num_waves:  INTEGER*4 number of arriving waves.
0015      c          noises:    INTEGER*4 number of (nominally) independent noise
0016      c                   sources used in the ESTIMATION model.
0017      c          anoises:   INTEGER*4 number of (nominally) independent noise
0018      c                   sources used in sample GENERATION.
0019      c          which:     INTEGER*4 selector for using "model" (which > 0) or
0020      c                   "actual" (which = 0) noise for
0021      c                   estimation.
0022      c          num_smp:   INTEGER*4 number of samples to process (in the
0023      c                   current experiment).
0024      c
0025      c-----
0026      c
0027      c          implicit none
0028      c          intrinsic index
0029      c          external caps, cap, nonblank
0030      c
0031      c          logical nonblank
0032      c          character*1 cap
0033      c
0034      c          integer*4 num_ants, num_waves, noises, anoises, which
0035      c          integer*4 num_smp
0036      c
0037      c          integer*4 linenum, it1, it2
0038      c          character*5 ianswer
0039      c
0040      10010 continue
0041      c          assign 10010 to linenum
0042      c          type 90010, ' Number of antennae (', num_ants, ')? '
0043      c          accept 90020, ianswer
0044      c          if ( nonblank( ianswer ) ) then
0045      c             read (unit=ianswer,fmt=*,err=20000) it1
0046      c             if ( it1 .le. 0 ) go to 20050
0047      c             num_ants = it1
0048      c          end if
0049      c
0050      10020 continue
0051      c          assign 10020 to linenum
0052      c          type 90010, ' Number of arriving waves (', num_waves, ')? '
0053      c          accept 90020, ianswer
0054      c          if ( nonblank( ianswer ) ) then
0055      c             read (unit=ianswer,fmt=*,err=20000) it1
0056      c             if ( it1 .le. 0 ) go to 20050
0057      c             num_waves = it1

```

GET_MUSIC_DEF

18-Apr-1988 14:18:06
25-Feb-1988 18:45:56VAX
GET

```

0058             end if
0059     c
0060     10025  continue
0061             assign 10025 to linenum
0062             type 90010, ' Number of noise generators for sample GENERATION (' ,
0063             1                               noises, ')? '
0064             accept 90020, ianswer
0065             if ( nonblank( ianswer ) ) then
0066                 read (unit=ianswer,fmt=*,err=20000) it1
0067                 if ( it1 .le. 0 ) go to 20050
0068                 noises = it1
0069             end if
0070     c
0071     10120  continue
0072             if ( which .gt. 0 ) then
0073                 ianswer = 'A/[M]'
0074             else
0075                 ianswer = '[A]/M'
0076             end if
0077             type 90030, ' Use actual or (different) model noise samples '
0078             1                               // 'for ESTIMATION (' // ianswer // ')? '
0079             accept 90020, ianswer
0080             if ( nonblank( ianswer ) ) then
0081                 call caps( ianswer )
0082                 it1 = index( ianswer, 'A' )
0083                 it2 = index( ianswer, 'M' )
0084                 if ( ( it1 .eq. 0 ) .and. ( it2 .eq. 0 ) ) go to 10120
0085                 if ( it1 .gt. it2 ) then
0086                     which = 0
0087                 else
0088                     which = 1
0089                 end if
0090             end if
0091     c
0092             if ( which .gt. 0 ) then
0093     10125  continue
0094             assign 10125 to linenum
0095             type 90010, ' Number of noise generators for ESTIMATION '
0096             1                               // 'noise model (' , noises, ')? '
0097             accept 90020, ianswer
0098             if ( nonblank( ianswer ) ) then
0099                 read (unit=ianswer,fmt=*,err=20000) it1
0100                 if ( it1 .le. 0 ) go to 20050
0101                 noises = it1
0102             end if
0103             else
0104                 noises = anoises
0105             end if
0106     c
0107     10030  continue
0108             assign 10030 to linenum
0109             type 90010, ' Number of samples per trial to process (' , num_smp,
0110             1                               ')? '
0111             accept 90020, ianswer
0112             if ( nonblank( ianswer ) ) then
0113                 read (unit=ianswer,fmt=*,err=20000) it1
0114                 if ( it1 .le. 0 ) go to 20050

```

GET_MUSIC_DEF

18-Apr-1988 14:18:06

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```

0115          num_smp = it1
0116          end if
0117      c
0118          type *
0119          type 90030, ' Above values OK (Y/N)? '
0120          accept 90020, ianswer
0121          if ( cap( ianswer( 1 : 1 ) ) .eq. 'N' ) go to 10010
0122      c
0123          return
0124      c
0125      20000  continue
0126          type *, char( 7 ), 'Error on input. Please re-enter.'
0127          go to linenum
0128      c
0129      20050  continue
0130          type *, char( 7 ), 'Must be > 0. Please re-enter.'
0131          go to linenum
0132      c
0133      90010  format ( $ a, i5, a )
0134      90020  format ( a5 )
0135      90030  format ( $ a )
0136      c
0137          end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	1196	PIC CON REL LCL SHR EXE RD NOWF
1 \$PDATA	390	PIC CON REL LCL SHR NOEXE RD NOWF
2 \$LOCAL	184	PIC CON REL LCL NOSHR NOEXE RD WF
Total Space Allocated	1760	

ENTRY POINTS

Address	Type	Name
0-00000000		GET_MUSIC_DEF

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type	Name
AP-00000010@	I*4	ANSISES	2-00000000	CHAR	ANSWER	2-00000008	I*4	ANSWER
**	I*4	LINENUM	AP-0000000C@	I*4	NOISES	AP-00000004@	I*4	NOISES
AP-00000008@	I*4	NUM_WAVES	AP-00000014@	I*4	WHICH			

13-Apr-1988 14:18:26
25-Aug-1987 14:27:25

```

0001      subroutine get_w_half( resid, size, p, eps, norm, w )
0002      c
0003      c-----
0004      c
0005      c          This subroutine computes the SQUARE ROOT OF the weighting
0006      c "matrix" (W) for the IRLS algorithm. W is a diagonal matrix, so the
0007      c result is returned in the VECTOR w.
0008      c
0009      c Parameters:
0010      c-----
0011      c          resid: size-element COMPLEX*16 vector of residuals for which
0012      c                  the weight matrix is to be computed.
0013      c          size:  INTEGER*4 number of residual elements (problem order).
0014      c          p:    REAL*4 value of power of residuals to minimize. That is
0015      c                  an Lp minimization is to be performed.
0016      c          eps:  REAL*8 value of minimum residual magnitude to be used
0017      c                  (see below).
0018      c          norm: REAL*8 value to which to normalize the SQUARE ROOT OF
0019      c                  the maximum weight. If norm (= 0.0d0)
0020      c                  no normalization occurs. Otherwise,
0021      c                  all weight SQUARE ROOT values (w(i))
0022      c                  are divided by a value which results
0023      c                  a maximum weight SQUARE ROOT of norm.
0024      c          w:    size-element REAL*8 vector of the SQUARE ROOTS OF the
0025      c                  computed weights.
0026      c
0027      c Operation:
0028      c-----
0029      c The elements of w are computed as follows:
0030      c          | resid(i) | ^ ( ( p - 2.0 ) / 2.0 )   if | resid(i) | > eps
0031      c          | eps | ^ ( ( p - 2.0 ) / 2.0 )       otherwise.
0032      c If norm > 0.0d0, w(i) is then divided by norm / ( max w(i) ).
0033      c
0034      c-----
0035      c
0036      c implicit none
0037      c intrinsic cdabs
0038      c
0039      c integer*4 size
0040      c real*4 p
0041      c real*8 eps, norm, w( size )
0042      c complex*16 resid( size )
0043      c
0044      c integer*4 i
0045      c real*4 pm2
0046      c real*8 epm2, wmult
0047      c
0048      c pm2 = 0.5 * p - 1.0
0049      c epm2 = eps ** pm2
0050      c wmult = -1.0d0
0051      c
0052      c do i = 1, size
0053      c     w(i) = cdabs( resid(i) )
0054      c     if ( w(i) .lt. eps ) then
0055      c       w(i) = epm2
0056      c     else
0057      c       w(i) = w(i) ** pm2

```

GET_W_HALF

18-Apr-1988 14:18:26
25-Aug-1987 14:27:25

```

0058         end if
0059         if ( norm .gt. 0.0d0 ) then
0060             if ( w(i) .gt. wmult ) wmult = w(i)
0061         end if
0062     end do
0063     c
0064         if ( norm .gt. 0.0d0 ) then
0065             wmult = norm / wmult
0066             do i = 1, size
0067                 w(i) = w(i) * wmult
0068             end do
0069         end if
0070     c
0071     return
0072     end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	228	PIC CON REL LCL SHR EXE RD NOV
2 \$LOCAL	96	PIC CON REL LCL NOSHR NOEXE RD V
Total Space Allocated	324	

ENTRY POINTS

Address	Type	Name
0-00000000		GET_W_HALF

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type
**	R*8	EPM2	AP-00000010@	R*8	EPS	**	I*4
AP-0000000C@	R*4	P	**	R*4	PM2	AP-00000008@	I*4

ARRAYS

Address	Type	Name	Bytes	Dimensions
AP-00000004@	C*16	RESID	**	(*)
AP-00000018@	R*8	W	**	(*)

18-Apr-1988 14:19:09
19-Dec-1987 09:14:31

```

0001      subroutine make_noise( out, n, defs, pars, corr, size )
0002      c
0003      c -----
0004      c
0005      c           This subroutine computes a COMPLEX*16 noise vector.
0006      c
0007      c Parameters:
0008      c -----
0009      c           out = COMPLEX*16 n-element output noise vector.
0010      c           n = INTEGER*4 number of elements in the output vector.
0011      c           defs = INTEGER*4 size x 2 matrix whose first column is the
0012      c                   vector of noise source types (see subroutine
0013      c                   READ_NOISE), and whose second column is the
0014      c                   vector of counters for intermittent and burst
0015      c                   noise.
0016      c           pars = REAL*4 size x 3 matrix of noise source parameters (see
0017      c                   subroutine READ_NOISE).
0018      c           corr = REAL*8 n x size matrix of noise source multipliers for
0019      c                   each output element ("correlation" matrix).
0020      c           size = INTEGER*4 number of noise sources in use.
0021      c
0022      c           NOTE: Intermittent and burst noise sources are controlled by
0023      c                   the parameter defs, above. If the defs(*,2) entry for an intermittent
0024      c                   or burst source is 1, the source is activated for the current output
0025      c                   vector. All intermittent and burst source counts are decremented aft
0026      c                   output generation, and sources whose counts reach zero are then
0027      c                   supplied with new counts (based on the appropriate entries in pars).
0028      c
0029      c -----
0030      c
0031      c           implicit none
0032      c           intrinsic dshft
0033      c           external open_va, free_va, exit_va
0034      c           external c8_noise, noise_combine, uni
0035      c           include '6000$RT:[CHUCK.RESEARCH.FORTDIRJVATYPES.TXT'
0036      c
0037      c           real*4 uni
0038      c
0039      c           integer*4 n, size, defs( size, 1 )
0040      c           real*4 pars( size, 1 )
0041      c           real*8 corr( n, size )
0042      c           complex*16 out( n )
0043      c
0044      c           integer*4 linear( 1 ), va_status, source_noise_add, i
0045      c           character*80 va_msg
0046      c
0047      c           linear(1) = size
0048      c
0049      c           Get space for source noise vector, then generate size COMPLEX
0050      c           noise values in the vector.
0051      c
0052      c           call open_va( source_noise_add, linear, 1, sc_type, va_status, va_msg )
0053      c           call exit_va( va_status, va_msg )
0054      c           call c8_noise( %val( source_noise_add ), size, defs, pars )
0055      c
0056      c           Now multiply the result noise source vector by corr (see
0057      c           subroutine NOISE_COMBINE), and discard the space used by the source
0058      c
0059      c
0060      c
0061      c
0062      c
0063      c
0064      c
0065      c
0066      c
0067      c
0068      c

```

MAKE_NOISE

18-Apr-1988 14:1

19-Dec-1987 09:1

```

0069      c      noise vector.
0070      c
0071      call noise_combine( corr, n, size, %val( source_noise_add ), def
0072      1
0073      call free_va( source_noise_add, va_status, va_msg )
0074      call exit_va( va_status, va_msg )
0075      c
0076      c      Finally, re-trigger intermittent and burst noise sources
0077      c      appropriate.
0078      c
0079      do i = 1, size
0080      if ( jshft( defs(i,1), -16 ) .ne. 0 ) then
0081      defs(i,2) = defs(i,2) - 1
0082      if ( defs(i,2) .le. 0 ) then
0083      defs(i,2) = pars(i,3) * ( 0.5 + uni( 10571 ) ) + 0.5
0084      end if
0085      end if
0086      end do
0087      c
0088      return
0089      end.

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	316	PIC CON REL LCL SHR EXE R
1 \$PDATA	12	PIC CON REL LCL SHR NOEXE R
2 \$LOCAL	412	PIC CON REL LCL NOSHR NOEXE R
Total Space Allocated	740	

ENTRY POINTS

Address	Type	Name
0-00000000		MAKE_NOISE

VARIABLES

Address	Type	Name	Address	T
**	I*4	I	AP-00000000	
AP-00000018	I*4	SIZE	2-00000053	
2-00000004	CHAR	VA_MSG	2-00000054	

18-Apr-1988 14:20:47

9-Sep-1987 17:14:46

```

0001      complex*16 function modulated( wave, num_waves, carr_camp, wpars,
0002      1
0003      c
0004      c-----
0005      c
0006      c          This function returns the complex amplitude of wave number
0007      c          wave based on the wave's modulation characteristics and carrier
0008      c          amplitude and phase. That is, this routine returns the wave-th elem
0009      c          of the "signal-in-space" vector at sample number samp_num.
0010      c
0011      c          Parameters:
0012      c          -----
0013      c          wave:  INTEGER*4 number of the wave for which to compute the
0014      c                   complex amplitude.
0015      c          num_waves:  INTEGER*4 number of waves present.
0016      c          carr_camp:  num_waves-element COMPLEX*16 vector of carrier
0017      c                   complex amplitudes (from SNRs and "starting"
0018      c                   phases).
0019      c          wpars = num_waves x 9 REAL*8 matrix whose columns include:
0020      c                   4) vector of modulation cycle lengths, in
0021      c                      number of carrier cycles per modulat
0022      c                      cycle.
0023      c                   5) vector of modulation duty cycles (expressed
0024      c                      as fractions of modulation cycle
0025      c                      lengths, 0.0 < mod_duty(i) (<= 1.0)
0026      c                      (apply only to ramp and rectangular
0027      c                      modulation).
0028      c                   6) vector of "starting" modulation phases, in
0029      c                      number of carrier cycles.
0030      c                   7) vector of amplitude modulation ratios
0031      c                      (maximum / minimum).
0032      c                   8) vector of phase modulation total phase
0033      c                      shifts (in RADIANS).
0034      c          mtypes = num_waves x 2 INTEGER*4 matrix whose first column is
0035      c                   the vector of amplitude modulation types and
0036      c                   whose second column is the vector of phase
0037      c                   modulation types (see below).
0038      c          samp_num:  INTEGER*4 number of the sample for which the comp
0039      c                   amplitude is to be computed.
0040      c          samp_int:  REAL*8 number of carrier cycles between samples.
0041      c
0042      c          Modulation types:  1 = sine, 2 = triangle, 3 = increasing ramp,
0043      c                   4 = decreasing ramp, 5 = rectangular.
0044      c
0045      c-----
0046      c
0047      c          implicit none
0048      c          intrinsic dcmplx, dcos, dsin
0049      c          external mod_val, mod_phase
0050      c
0051      c          real*8 mod_val, mod_phase
0052      c
0053      c          integer*4 wave, num_waves, samp_num, mtypes( num_waves, 1 )
0054      c          real*8 wpars( num_waves, 1 ), samp_int
0055      c          complex*16 carr_camp( num_waves )
0056      c
0057      c          real*8 temp, mod_phs

```

MODULATED 18-Apr-1988 14:20:47 V/
9-Sep-1987 17:14:46 M(

```

0058 c      mod_phs = mod_phase( wave, num_waves, wpars, samp_num, samp_int )
0059
0060 c
0061      temp = mod_val( mtypes(wave,1), mod_phs, wpars(wave,5) )
0062      modulated = carr_camp(wave) *
0063      1      ( temp + ( ( 1.0d0 - temp ) / wpars(wave,7) ) )
0064 c
0065      temp = 2.0d0 * wpars(wave,8) *
0066      1      ( mod_val( mtypes(wave,2), mod_phs, wpars(wave,5) ) - 0.5d0 )
0067      modulated = modulated * dcmplx( dcos( temp ), dsin( temp ) )
0068 c
0069      return
0070      end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 %CODE	308	PIC CON REL LCL SHR EXE RD NOWRT LI
2 %LOCAL	192	PIC CON REL LCL NOSHR NOEXE RD WRT QI
Total Space Allocated	500	

ENTRY POINTS

Address	Type	Name
0-00000000	C*16	MODULATED

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type	Name
2-00000000	R*8	MOD_PHS	AP-0000000C@	I*4	NUM_WAVES	AP-00000020@	R*8	SAMP
**	R*8	TEMP	AP-00000008@	I*4	WAVE			

ARRAYS

Address	Type	Name	Bytes	Dimensions
AP-00000010@	C*16	CARR_CAMP	**	(*)
AP-00000018@	I*4	MTYPES	**	(*, 1)
AP-00000014@	R*8	WPARS	**	(*, 1)

MOD_VAL

18-Apr-1988 14:20:47
9-Sep-1987 17:14:46

```

0108          if ( phs .le. duty ) then
0109              mod_val = 1.0d0 - ( phs / duty )
0110          else
0111              mod_val = 0.0d0
0112          end if
0113      c
0114          else
0115              if ( phs .le. duty ) then
0116                  mod_val = 1.0d0
0117              else
0118                  mod_val = 0.0d0
0119              end if
0120      c
0121          end if
0122          return
0123      c
0124          end

```

! Rectangular.

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	204	PIC CON REL LCL SHR EXE RD NOWR
2 \$LOCAL	24	PIC CON REL LCL NOSHR NOEXE RD WR
Total Space Allocated	228	

ENTRY POINTS

Address	Type	Name
0-00000000	R*8	MOD_VAL

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type	N
2-00000014	I*4	DMOD	AP-0000000C	R*8	DUTY	2-00000010	I*4	I
**	I*4	RMOD	2-00000008	I*4	SMOD	2-0000000C	I*4	T

FUNCTIONS AND SUBROUTINES REFERENCED

Type	Name
R*8	MTH\$DSIN

18-Apr-1988 14:21:33
18-Dec-1987 17:42:03

```

0001      subroutine music_init( waves, wpars, ccamp,
0002      1                          noises, ndefs, npars,
0003      2                          anoises, andefs, anpars, which )
0004      c
0005      c-----
0006      c
0007      c      This subroutine computes the complex amplitudes of the carriers
0008      c      of the waves incident on an antenna array, and initializes counters for
0009      c      use by intermittent noise generators.
0010      c
0011      c      Parameters:
0012      c      -----
0013      c      waves:  INTEGER*4 number of waves incident on the array.
0014      c      wpars:  waves x 9 REAL*8 matrix whose first column is the
0015      c      vector of incident wave strengths, in db
0016      c      (referred to an arbitrary unit amplitude), and
0017      c      whose second column is the vector of carrier
0018      c      starting phases, in RADIANS.
0019      c      ccamp:  waves-element COMPLEX*16 output vector of carrier
0020      c      complex amplitudes (from SNRs and "starting"
0021      c      phases). Expressed in "rectangular"
0022      c      coordinates (i.e.,  $A * \exp(j * \phi)$  ).
0023      c      noises:  INTEGER*4 number of noise generators in use in the
0024      c      ESTIMATION noise model.
0025      c      ndefs:  noises x 2 INTEGER*4 matrix whose first column is the
0026      c      vector of noise types (see subroutine
0027      c      READ_NOISE), and whose second column is the
0028      c      vector of counts-to-output for intermittent
0029      c      noise generators. This matrix applies to the
0030      c      ESTIMATION noise model.
0031      c      npars:  noises x 3 REAL*4 array of noise parameters (see
0032      c      subroutine READ_NOISE). This matrix applies
0033      c      to the ESTIMATION noise model.
0034      c      anoises:  INTEGER*4 number of noise generators in use for
0035      c      ACTUAL SAMPLE GENERATION.
0036      c      andefs:  anoises x 2 INTEGER*4 matrix whose first column is the
0037      c      vector of noise types (see subroutine
0038      c      READ_NOISE), and whose second column is the
0039      c      vector of counts-to-output for intermittent
0040      c      noise generators. This matrix applies to
0041      c      ACTUAL SAMPLE GENERATION.
0042      c      anpars:  anoises x 3 REAL*4 array of noise parameters (see
0043      c      subroutine READ_NOISE). This matrix applies
0044      c      to ACTUAL SAMPLE GENERATION.
0045      c      which:  INTEGER*4 value which determines whether the actual
0046      c      noise samples used in generating the
0047      c      observations are to be used in the estimation
0048      c      routines (which = 0), or a separate estimation
0049      c      noise model is to be used (which > 0). Note
0050      c      that if which is > 0 then anoises, andefs, and
0051      c      anpars are used; otherwise only noises, ndefs,
0052      c      and npars are used.
0053      c
0054      c-----
0055      c
0056      implicit none
0057      intrinsic dcplx, jshft

```


MUSIC_INIT

18-Apr-1988. 14:21:38

18-Dec-1987 17:42:08

```

0058      external rect, uni
0059      c
0060      real*4 uni
0061      complex*16 rect
0062      c
0063      integer*4 waves, noises, anoises, which
0064      integer*4 ndefs( noises, 1 ), andefs( anoises, 1 )
0065      real*4 npars( noises, 1 ), anpars( anoises, 1 )
0066      real*8 wpars( waves, 1 )
0067      complex*16 ccamp( waves )
0068      c
0069      integer*4 i, it1
0070      c
0071      do i = 1, waves
0072          ccamp(i) = rect( dcmplx( 10.0d0 ** ( wpars(i,1) / 20.0d0 ),
0073              1
0074              wpars(i,2) )
0075          end do
0076      c
0077      do i = 1, noises
0078          it1 = jshft( ndefs(i,1), -16 )
0079          if ( it1 .ne. 0 ) then
0080              ndefs(i,2) = npars(i,3) * ( 0.5 + uni( 10571 ) ) + 0.5
0081          else
0082              ndefs(i,2) = 0
0083          end if
0084      end do
0085      c
0086      if ( which .gt. 0 ) then
0087          do i = 1, anoises
0088              it1 = jshft( andefs(i,1), -16 )
0089              if ( it1 .ne. 0 ) then
0090                  andefs(i,2) = anpars(i,3) * ( 0.5 + uni( 10571 ) ) + 0.5
0091              else
0092                  andefs(i,2) = 0
0093              end if
0094          end do
0095      end if
0096      c
0097      return
0098      end

```

18-Apr-1988 14:21:58
17-Nov-1987 07:58:22

```

0001      subroutine music_l2p( a, ants, waves, rx, rw, fuzz, p, info )
0002      c
0003      c -----
0004      c
0005      c           This subroutine computes the L2 estimate of the P matrix, giv
0006      c           the (L2 estimate of the) A matrix (from the MUSIC algorithm).
0007      c
0008      c           Paramters:
0009      c           -----
0010      c           a:  ants x waves COMPLEX*16 (L2) estimate of the A matrix.
0011      c           ants:  INTEGER*4 number of antenna array elements.
0012      c           waves:  INTEGER*4 number of impinging waves.
0013      c           rx:  ants x ants COMPLEX*16 Rx matrix estimate (signal
0014      c                   covariance matrix).
0015      c           rw:  ants x ants COMPLEX*16 Rw matrix estimate (noise
0016      c                   covariance matrix).  NOTE tha
0017      c                   rw must NOT be normalized by
0018      c                   lambda-min.
0019      c           fuzz:  REAL*8 value to be taken for zero ("tolerance").
0020      c           p:  waves x waves COMPLEX*16 output P matrix estimate (source
0021      c                   covariance matrix).
0022      c           info:  INTEGER*4 status indicator (output); 0 for successful
0023      c                   estimation, non-0 on failure c
0024      c                   singular value decomposition.
0025      c
0026      c -----
0027      c
0028      c           implicit none
0029      c           external open_va, free_va, exit_va
0030      c           external dc_mppsinv, sqdc_matdif, dc_matmpy, dc_matmpyherm
0031      c
0032      c           integer*4 ants, waves, info
0033      c           real*8 fuzz
0034      c           complex*16 a( ants, waves ), rx( ants, ants ), rw( ants, ants )
0035      c           complex*16 p( waves, waves )
0036      c
0037      c           integer*4 dc_type//, status
0038      c           integer*4 ants_sqr( 2 ), waves_by_ants( 2 )
0039      c           integer*4 prod_add, diff_add, wk1_add
0040      c           character*80 msg
0041      c
0042      c           ants_sqr(1)      = ants
0043      c           ants_sqr(2)      = ants
0044      c           waves_by_ants(1) = waves
0045      c           waves_by_ants(2) = ants
0046      c
0047      c           call open_va( prod_add, waves_by_ants, 2, dc_type, status, msg )
0048      c           call exit_va( status, msg )
0049      c           call open_va( diff_add, ants_sqr, 2, dc_type, status, msg )
0050      c           call exit_va( status, msg )
0051      c           call open_va( wk1_add, waves_by_ants, 2, dc_type, status, msg )
0052      c           call exit_va( status, msg )
0053      c
0054      c           call dc_mppsinv( a, ants, waves, fuzz, %val( prod_add ), info )
0055      c           call sqdc_matdif( rx, rw, ants, %val( diff_add ) )
0056      c           call dc_matmpy( %val( prod_add ), waves, ants, %val( diff_add ), ants,
0057      c                   1, %val( wk1_add ) )

```

MUSIC_L2P

18-Apr-1988 14:21:5

17-Nov-1987 07:58:2

```

0058      call dc_matpyherm( %val( wk1_add ), waves, ants,
0059      1                      %val( prod_add ), waves, p )
0060      c
0061      call free_va( prod_add, status, msg )
0062      call exit_va( status, msg )
0063      call free_va( diff_add, status, msg )
0064      call exit_va( status, msg )
0065      call free_va( wk1_add, status, msg )
0066      call exit_va( status, msg )
0067      c
0068      return
0069      end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	414	PIC CON REL LCL SHR EXE RD N
1 \$PDATA	4	PIC CON REL LCL SHR NOEXE RD N
2 \$LOCAL	524	PIC CON REL LCL NOSHR NOEXE RD
Total Space Allocated	942	

ENTRY POINTS

Address	Type	Name
0-00000000		MUSIC_L2P

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type
AP-00000008	1*4	ANTS	2-00000060	1*4	DC_TYPE	2-0000006C	1*4
AP-00000020	1*4	INFD	2-00000010		CHAR MSG	2-00000068	1*4
AP-0000000C	1*4	WAVES	2-00000070	1*4	WK1_ADD		

ARRAYS

Address	Type	Name	Bytes	Dimensions
AP-00000004	C*16	A	**	(*, *)
2-00000000	1*4	ANTS_SQR	8	(2)
AP-0000001C	C*16	P	**	(*, *)
AP-00000014	C*16	RJ	**	(*, *)
AP-00000010	C*16	RX	**	(*, *)
2-00000008	1*4	WAVES_BY_ANTS	8	(2)

18-Apr-1988 14:3
16-Mar-1988 09:3

```

0001      subroutine music_lpp_each( obsmat, samps, ants, a, waves, p, eps
0002      1                               norm, tol, pvm, resm, mxitlim,
0003      2                               mnitlim, atitlim, min_de
0004      c
0005      c-----
0006      c
0007      c      This subroutine forms Lp estimates of each of the signal
0008      c      vector samples which produce a collection of observation vector
0009      c      from an antenna array. Estimates are formed using the IRLS algo
0010      c
0011      c      Parameters:
0012      c      -----
0013      c      obsmat:  samps x ants COMPLEX*16 matrix whose ROWS are t
0014      c                  noisy observations from the antenna arra
0015      c      samps:  INTEGER*4 number of observation samples.
0016      c      ants:  INTEGER*4 number of elements in the antenna array
0017      c      a:    ants x waves COMPLEX*16 phase shift matrix for the s
0018      c                  array and the arriving waves (the column
0019      c                  are the "signal vectors").
0020      c      waves:  INTEGER*4 number of arriving waves.
0021      c      p:    REAL*4 power of the residuals to be minimized (Lp
0022      c                  estimation to be used).
0023      c      eps:   REAL*8 minimum residual magnitude to be used.
0024      c      norm:  REAL*8 value to which to normalize the SQUARE ROC
0025      c                  the maximum IRLS weight. If norm <= 0.0
0026      c                  normalization occurs. Otherwise, all we
0027      c                  SQUARE ROOT values (w(i)) are divided by
0028      c                  value which results in a maximum weight
0029      c                  ROOT of norm.
0030      c      tol:   REAL*8 value to be taken for zero ("tolerance").
0031      c      pvm:   waves x samps COMPLEX*16 output matrix whose COLU
0032      c                  the estimates of the signal-in-space vec
0033      c                  samples.
0034      c      resm:   samps x ants COMPLEX*16 output matrix whose ROWS
0035      c                  residuals from the above estimates.
0036      c      mxitlim:  INTEGER*4 number of iterations to permit for t
0037      c                  algorithm.
0038      c      mnitlim:  INTEGER*4 minimum number to which the iteratic
0039      c                  counter was decremented by the IRLS algo
0040      c      Note:  mxitlim - mnitlim is usually the
0041      c                  number of iterations used in any IRLS es
0042      c                  however, since routine DCV_IRLS can plac
0043      c                  NEGATIVE value in mnitlim, this possibil
0044      c                  should be checked.
0045      c      atitlim:  INTEGER*4 number of the sample to which mnitli
0046      c                  applies.
0047      c      min_del:  REAL*8 square of the norm of the difference be
0048      c                  two subsequent estimates of the same vec
0049      c                  This variable is set to the difference p
0050      c                  by subroutine DCV_IRLS for sample number
0051      c                  atitlim.
0052      c
0053      c-----
0054      c
0055      c      implicit none
0056      c      external open_va, free_va, exit_va
0057      c      external c16_vec_ext, dcv_irls, c16_col_ins, c16_vec_ins

```

MUSIC_LPP_EACH

18-Apr-1988 14:22:20

16-Mar-1988 09:26:50

```

0058      c
0059          integer*4 samps, ants, waves, mxitlim, mnitlim, atitlim
0060          real*4 p
0061          real*8 eps, norm, tol, min_del
0062          complex*16 obsmat( samps, ants ), a( ants, waves )
0063          complex*16 pvm( waves, samps ), resm( samps, ants )
0064      c
0065          integer*4 status, obs_add, sspc_add, resid_add
0066          integer*4 ants_lin( 1 ), waves_lin( 1 )
0067          integer*4 di_type/3/, sr_type/4/, dr_type/5/, dc_type/7/
0068          integer*4 i, itlim
0069          real*8 del
0070          character*80 msg
0071      c
0072          ants_lin(1) = ants
0073          waves_lin(1) = waves
0074          call open_va( obs_add, ants_lin, 1, dc_type, status, msg )
0075          call exit_va( status, msg )
0076          call open_va( sspc_add, waves_lin, 1, dc_type, status, msg )
0077          call exit_va( status, msg )
0078          call open_va( resid_add, ants_lin, 1, dc_type, status, msg )
0079          call exit_va( status, msg )
0080      c
0081          min_del = 0.0d0
0082          mnitlim = mxitlim
0083          atitlim = 0
0084          do i = 1, samps
0085              itlim = mxitlim
0086              call c16_vec_ext( obsmat, samps, ants, i, %val( obs_add ) )
0087              call dcw_irls( a, ants, waves, %val( obs_add ),
0088                  1          p, eps, norm, tol, itlim,
0089                  2          %val( sspc_add ), %val( resid_add ), del )
0090              if ( itlim .lt. mnitlim ) then
0091                  mnitlim = itlim
0092                  atitlim = i
0093                  min_del = del
0094              end if
0095              call c16_col_ins( %val( sspc_add ), waves, pvm, waves, samps, i )
0096              call c16_vec_ins( %val( resid_add ), ants, resm, samps, ants, i )
0097          end do
0098      c
0099          call free_va( obs_add, status, msg )
0100          call exit_va( status, msg )
0101          call free_va( sspc_add, status, msg )
0102          call exit_va( status, msg )
0103          call free_va( resid_add, status, msg )
0104          call exit_va( status, msg )
0105      c
0106          return
0107          end

```

18-Apr-1988 14:22:44 (

17-Nov-1987 08:36:13 (

```

0001      subroutine music_lpp_speis( a, n, m, w, robs, rnoise,
0002      1
0003      fuzz, pmat, info )
0004      c-----
0005      c
0006      c          This subroutine computes an Lp estimate of the P matrix, given
0007      c          the (L2 estimate of the) A matrix (from the MUSIC algorithm). The
0008      c          approach used is due to Speiser, as follows:
0009      c          pmat = ( a* w a )- a* w ( robs - rnoise ) w a ( a* w a )-,
0010      c          where * signifies conjugate transpose and ( )- is matrix inverse.
0011      c
0012      c          Paranters:
0013      c          -----
0014      c          a:  n x m COMPLEX*16 (L2) estimate of the A matrix.
0015      c          n:  INTEGER*4 number antenna array elements.
0016      c          m:  INTEGER*4 number of impinging waves.
0017      c          w:  n-element REAL*8 vector containing the (non-zero) elements
0018      c          of the IRLS weight matrix (from MUSIC_LPW_YAR,
0019      c          usually).
0020      c          robs: n x n COMPLEX*16 Rx matrix estimate (signal covariance
0021      c          matrix).
0022      c          rnoise: n x n COMPLEX*16 Rw matrix estimate (noise covariance
0023      c          matrix). NOTE that rnoise must NOT be
0024      c          normalized by lambda-min.
0025      c          fuzz: REAL*8 value to be taken for zero ("tolerance").
0026      c          pmat: m x m COMPLEX*16 output P matrix estimate (source
0027      c          covariance matrix).
0028      c          info: INTEGER*4 status indicator (output); 0 for successful
0029      c          estimation, non-0 on failure of singular value
0030      c          decomposition.
0031      c-----
0032      c
0033      c          implicit none
0034      c          external open_va, free_va, exit_va
0035      c          external dc_mppsinv, sqdc_matdif
0036      c          external drdi_dc_mmpy, dc_drdi_mmpy, dc_matmpy, dc_matmpyherm
0037      c
0038      c          integer*4 n, m, info
0039      c          real*8 fuzz, w( n )
0040      c          complex*16 a( n, m ), robs( n, n ), rnoise( n, n ), pmat( m, m )
0041      c
0042      c          integer*4 dc_type/7/, status
0043      c          integer*4 n_sqr( 2 ), m_by_n( 2 ), n_by_m( 2 )
0044      c          integer*4 prod_add, diff_add, wk1_add, wk2_add
0045      c          character*80 msg
0046      c
0047      c          n_sqr(1) = n
0048      c          n_sqr(2) = n
0049      c          n_by_m(1) = n
0050      c          n_by_m(2) = m
0051      c          m_by_n(1) = m
0052      c          m_by_n(2) = n
0053      c          call open_va( prod_add, m_by_n, 2, dc_type, status, msg )
0054      c          call exit_va( status, msg )
0055      c          call open_va( diff_add, n_sqr, 2, dc_type, status, msg )
0056      c          call exit_va( status, msg )
0057

```

MUSIC_LPP_SPEIS

18-Apr-1988 14:22:44
17-Nov-1987 08:36:13

```

0058      call open_va( wk1_add, n_by_m, 2, dc_type, status, msg )
0059      call exit_va( status, msg )
0060      call open_va( wk2_add, m_by_n, 2, dc_type, status, msg )
0061      call exit_va( status, msg )
0062      c
0063      call drdi_dc_mmpy( w, n, a, m, %val( wk1_add ) )
0064      call dc_mppsinv( %val( wk1_add ), n, m, fuzz, %val( wk2_add ), info )
0065      call dc_drdi_mmpy( %val( wk2_add ), m, n, w, %val( prod_add ) )
0066      c
0067      call sqdc_matdif( robs, rnoise, n, %val( diff_add ) )
0068      call dc_matmpy( %val( prod_add ), m, n, %val( diff_add ), n,
0069      1                                     %val( wk2_add ) )
0070      call dc_matmpyherm( %val( wk2_add ), m, n, %val( prod_add ), m, pmat
0071      c
0072      call free_va( prod_add, status, msg )
0073      call exit_va( status, msg )
0074      call free_va( diff_add, status, msg )
0075      call exit_va( status, msg )
0076      call free_va( wk1_add, status, msg )
0077      call exit_va( status, msg )
0078      call free_va( wk2_add, status, msg )
0079      call exit_va( status, msg )
0080      c
0081      return
0082      end
    
```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	561	PIC CON REL LCL SHR EXE RD NOW
1 \$PDATA	4	PIC CON REL LCL SHR NOEXE RD NOW
2 \$LOCAL	644	PIC CON REL LCL NOSHR NOEXE RD W
Total Space Allocated		1209

ENTRY POINTS

Address	Type	Name
0-00000000		MUSIC_LPP_SPEIS

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type
2-00000068	1*4	DC_TYPE	2-00000074	1*4	DIFF_ADD	AP-0000001C@	R*8
AP-0000000C@	1*4	M	2-00000018		CHAR MSG	AP-00000008@	1*4
2-0000006C	1*4	STATUS	2-00000078	1*4	WK1_ADD	2-0000007C	1*4

19-Apr-1988 14:23:03
16-Mar-1988 10:16:26

```

0001      subroutine music_lpw_yar( obs, samps, ants, a, waves, p, eps,
0002      1                               norm, tol, weights, sspace,
0003      2                               resids, itlim, minit, del, wtfun )
0004      c
0005      c-----
0006      c
0007      c      This subroutine forms a "combined" Lp weight matrix and uses
0008      c      it to compute estimates of each of the signal vector samples which
0009      c      produce a collection of observation vector samples from an antenna
0010      c      array. The weight matrix is formed via the IRLS algorithm using the
0011      c      weight function selected by the INTEGER*4 value wtfun. If wtfun = 1,
0012      c      the WEIGHT function used is one over the p-th root of the sum (across
0013      c      the samples) of the residuals raised to the p-th power (I have labelled
0014      c      this approach the "Yariagadda" algorithm). If wtfun = 2, the LOSS
0015      c      function used (to compute the weight function) is the p-th root of the
0016      c      sum (across the samples) of the residuals raised to the p-th power (I
0017      c      have labelled this the "sum of powers loss" approach). NOTE that the
0018      c      actual "weight" matrix returned is composed of the SQUARE ROOTS of the
0019      c      computed weights.
0020      c
0021      c      Parameters:
0022      c      -----
0023      c      obs:  samps x ants COMPLEX*16 matrix whose ROWS are the
0024      c      noisy observations from the antenna array.
0025      c      samps:  INTEGER*4 number of observation samples.
0026      c      ants:  INTEGER*4 number of elements in the antenna array.
0027      c      a:    ants x waves COMPLEX*16 phase shift matrix for the antenna
0028      c      array and the arriving waves (the columns of a
0029      c      are the "signal vectors").
0030      c      waves:  INTEGER*4 number of arriving waves.
0031      c      p:    REAL*4 power of the residuals to be minimized (Lp
0032      c      estimation to be used).
0033      c      eps:  REAL*8 minimum residual magnitude to be used.
0034      c      norm:  REAL*8 value to which to normalize the SQUARE ROOT OF
0035      c      the maximum weight. If norm <= 0.0d0, no
0036      c      normalization occurs. Otherwise, all weight
0037      c      SQUARE ROOT values (w(i)) are divided by a
0038      c      value which results in a maximum weight SQUARE
0039      c      ROOT of norm.
0040      c      tol:  REAL*8 value to be taken for zero ("tolerance").
0041      c      weights:  ants-element REAL*8 vector of SQUARE ROOTS of the
0042      c      computed weights (diagonal elements of the
0043      c      weight matrix).
0044      c      sspace:  waves x samps COMPLEX*16 output matrix whose COLUMNS
0045      c      are the estimates of the signal-in-space vector
0046      c      samples.
0047      c      resids:  samps x ants COMPLEX*16 output matrix whose ROWS are
0048      c      the residuals from the above estimates.
0049      c      itlim:  INTEGER*4 number of iterations to permit for the IRLS
0050      c      algorithm. NOTE that calling this routine with
0051      c      itlim = 0 will result in a least squares (L2)
0052      c      estimate, regardless of the value of p.
0053      c      minit:  INTEGER*4 minimum number to which the iteration counter
0054      c      was decremented by the IRLS algorithm. Note:
0055      c      itlim - minit is ordinarily the number of
0056      c      iterations used in the IRLS estimate, however,
0057      c      if an error occurs in routine DC_MPPSINV

```



```

18-Apr-1988 14:23:1
16-Mar-1988 10:16:1

0058 c      ("info", below, returned with a non-zero
0059 c      value), this routine sets minit to the neg
0060 c      of its current value minus one, and exits.
0061 c      del: REAL*8 exit value of the square of the norm of the
0062 c      difference between subsequent weight vecto
0063 c      wtfun: INTEGER*4 value selecting whether the sum-of-power
0064 c      (wtfun = 1) or the sum-of-powers-loss (wtf
0065 c      2) method is to be used to compute the wei
0066 c      (diagonal elements of the weight matrix).
0067 c
0068 c-----
0069 c
0070 c      implicit none
0071 c      intrinsic cdabs
0072 c      external open_va, free_va, exit_va
0073 c      external dr_vecfill, dr_veccopy, drdi_dc_mmpy, dc_mppsinv
0074 c      external dc_drdrdi_mmpy, yar_est, yar_resids, dr_vdel_nmsq
0075 c
0076 c      real*8 dr_vdel_nmsq
0077 c
0078 c      integer*4 samps, ants, waves, itlim, minit, wtfun
0079 c      real*4 p
0080 c      real*8 eps, norm, tol, weights( ants ), del
0081 c      complex*16 obs( samps, ants ), a( ants, waves )
0082 c      complex*16 sspace( waves, samps ), resids( samps, ants )
0083 c
0084 c      integer*4 status, antslin( 1 ), antswaves( 2 ), wavesants( 2 ), i,
0085 c      integer*4 oldw_add, wa_add, psinv_add, info
0086 c      integer*4 dl_type/3/, sr_type/4/, dr_type/5/, dc_type/7/
0087 c      real*4 wexp, pinv
0088 c      real*8 maxweight, wnorm
0089 c      character*80 msg
0090 c
0091 c      antslin(1) = ants
0092 c      antswaves(1) = ants
0093 c      antswaves(2) = waves
0094 c      wavesants(1) = waves
0095 c      wavesants(2) = ants
0096 c      call open_va( oldw_add, antslin, 1, dr_type, status, msg )
0097 c      call exit_va( status, msg )
0098 c      call open_va( wa_add, antswaves, 2, dc_type, status, msg )
0099 c      call exit_va( status, msg )
0100 c      call open_va( psinv_add, wavesants, 2, dc_type, status, msg )
0101 c      call exit_va( status, msg )
0102 c
0103 c      pinv = 1.0 / p
0104 c      if ( wtfun .eq. 1 ) then                ! Sum of powers.
0105 c          wexp = - 0.5 * pinv
0106 c      else                                     ! Sum of powers loss.
0107 c          wexp = ( p / 2.0 ) - 1.0
0108 c      end if
0109 c      maxweight = eps ** wexp
0110 c
0111 c      Compute initial (L2) estimate and residuals therefor.
0112 c
0113 c      del = tol + .5
0114 c      call dr_vecfill( weights, ants, 1.0d0 )

```

MUSIC_LPW_YAR

18-Apr-1988 14:23:03

16-Mar-1988 10:16:26

```

0115      call dc_mppsinv( a, ants, waves, tol, %val( psinv_add ), info )
0116      if ( info .ne. 0 ) then
0117          minit = - itlim - 1
0118          go to 10000
0119      end if
0120      call yar_estst( obs, samps, ants, %val( psinv_add ), waves, sspace )
0121      call yar_resids( obs, samps, ants, a, waves, sspace, resids )
0122      c
0123      c          Iterate to the "Lp" solution, or exit if itlim = 0.
0124      c
0125      minit = itlim
0126      do while ( ( del .gt. tol ) .and. ( minit .gt. 0 ) )
0127          call dr_veccopy( weights, ants, %val( oldw_add ) )
0128      c
0129          do i = 1, ants
0130              weights(i) = 0.0d0
0131              do j = 1, samps
0132                  weights(i) = weights(i) + cdabs( resids(j,i) ) ** p
0133              end do
0134              if ( wtfun .eq. 2 ) weights(i) = weights(i) ** plnv
0135              if ( weights(i) .lt. eps ) then
0136                  weights(i) = maxweight
0137              else
0138                  weights(i) = weights(i) ** wexp
0139              end if
0140          end do
0141      c
0142          if ( norm .gt. 0.0d0 ) then
0143              wnorm = weights(1)
0144              do i = 2, ants
0145                  if ( weights(i) .gt. wnorm ) wnorm = weights(i)
0146              end do
0147              if ( wnorm .gt. 0.0d0 ) then
0148                  do i = 1, ants
0149                      weights(i) = weights(i) / wnorm
0150                  end do
0151              end if
0152          end if
0153      c
0154          call drdi_dc_mmpy( weights, ants, a, waves, %val( wa_add ) )
0155          call dc_mppsinv( %val( wa_add ), ants, waves, tol,
0156                        1 %val( psinv_add ), info )
0157          if ( info .ne. 0 ) then
0158              minit = - minit - 1
0159              go to 10000
0160          end if
0161          call dc_drdi_mmpy( %val( psinv_add ), waves, ants, weights,
0162                        1 %val( psinv_add ) )
0163          call yar_estst( obs, samps, ants, %val( psinv_add ), waves, sspace )
0164          call yar_resids( obs, samps, ants, a, waves, sspace, resids )
0165      c
0166          minit = minit - 1
0167          del = dr_vdel_nmsq( weights, %val( oldw_add ), ants, tol )
0168      end do
0169      c
0170      10000      continue
0171      call free_va( oldw_add, status, msg )

```

MUSIC_LPW_YAR

18-Apr-1988 14:23:03

16-Mar-1988 10:16:26

```

0172      call exit_va( status, msg )
0173      call free_va( wa_add, status, msg )
0174      call exit_va( status, msg )
0175      call free_va( psinv_add, status, msg )
0176      call exit_va( status, msg )
0177      c
0178      return
0179      end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	1079	PIC CON REL LCL SHR EXE RD NOWI
1 \$PDATA	16	PIC CON REL LCL SHR NOEXE RD NOWI
2 \$LOCAL	724	PIC CON REL LCL NOSHR NOEXE RD W
Total Space Allocated		1819

ENTRY POINTS

Address	Type	Name
0-00000000		MUSIC_LPW_YAR

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type	Name
AP-0000000C	I*4	ANTS	2-00000088	I*4	DC_TYPE	AP-0000003C	R*8	
2-00000084	I*4	DR_TYPE	AP-0000001C	R*8	EPS	**	I*4	
AP-00000034	I*4	ITLIM	**	I*4	J	2-00000068	R*8	
2-00000014	CHAR	MSG	AP-00000020	R*8	NORM	2-00000074	I*4	
**	R*4	PINV	2-0000007C	I*4	PSINV_ADD	AP-00000008	I*4	
2-00000070	I*4	STATUS	AP-00000024	R*8	TOL	AP-00000014	I*4	
**	R*4	WEXP	**	R*8	WNORM	AP-00000040	I*4	

ARRAYS

Address	Type	Name	Bytes	Dimensions
AP-00000010	C*16	A	**	(* , *)
2-00000000	I*4	ANTSLIN	4	(1)
2-00000004	I*4	ANTSWAVES	8	(2)
AP-00000004	C*16	OBS	**	(* , *)
AP-00000030	C*16	RESIDS	**	(* , *)
AP-0000002C	C*16	SSPACE	**	(* , *)
2-0000000C	I*4	WAVESANTS	8	(2)
AP-00000028	R*8	WEIGHTS	**	(* , *)

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```

0001      logical function
0002      1      music_plot_spect( apars, num_ants, wavelen, eigvec, select
0003      2      mi_begp, mi_delp, fskip, count, isf
0004      c
0005      -----
0006      c
0007      c      This function plots MUSIC "power" spectra according to the
0008      c      user's instructions. The function returns .true. if the operator
0009      c      enters X (to stop DOA spectrum plots) in response to the "do you w
0010      c      a replot?" prompt, and .false. otherwise.
0011      c
0012      c      Parameters:
0013      c      -----
0014      c      apars = matrix whose first column is the vector of ranges
0015      c      the origin) of array antenna elements and (
0016      c      second column is the vector of angles (fron
0017      c      "x" axis) of array antenna elements (in
0018      c      RADIANS). num_ants x 2. REAL*8.
0019      c      Note: the position of the ith antenna element is
0020      c      apars(i,1) * cos( apars(i,2) ) (x position),
0021      c      apars(i,1) * sin( apars(i,2) ) (y position).
0022      c      num_ants = number of elements in antenna array. INTEGER*4
0023      c      wavelen = wavelength of the incident waves. REAL*8. (Note
0024      c      that the units of wavelength must be the s;
0025      c      as those in which the array_range(*) are
0026      c      expressed.)
0027      c      eigvec: num_ants x num_ants COMPLEX*16 matrix whose column
0028      c      are the applicable generalized eigenvectors;
0029      c      The eigenvectors are ASSUMED to be arranged
0030      c      ascending eigenvalue order, with the
0031      c      eigenvector corresponding to the smallest
0032      c      eigenvalue occupying column #1, and so on.
0033      c      select: INTEGER*4 number of eigenvectors (columns of eigv
0034      c      to be used in the computation (number of
0035      c      "noise" eigenvectors).
0036      c      mi_begp: REAL*4 minimum permitted value of the power (p)
0037      c      which to raise the individual terms of the
0038      c      which makes up the reciprocal of a MUSIC DO
0039      c      spectrum point (p = 2 in "ordinary" MUSIC)
0040      c      mi_delp: REAL*4 minimum increment value for p.
0041      c      fskip: INTEGER*4 number of p values to skip before the fir
0042      c      value for which to plot.
0043      c      count: INTEGER*4 total number of p values for which to plo
0044      c      iskip: INTEGER*4 number of p values to skip between values
0045      c      which to plot.
0046      c
0047      c      -----
0048      c
0049      c      implicit none
0050      c      external music_spectrumx, music_spectrumy, vecext, log_scale
0051      c      external cap, r4_col_ins, r4_col_ext, value_at
0052      c      external open_va, exit_va, free_va
0053      c      external tek_start, tek_grid, tek_title, tek_plot, tek_vec2, tek_fi
0054      c      external tek_point, tek_label
0055      c      include '6000$RT:[CHUCK.RESEARCH.FORTDIR]VATYPES.TXT'
0056      c      include '(real_constants)'
0057      c
0058      c
0059      c
0060      c
0061      c
0062      c
0063      c
0064      c
0065      c
0066      c
0067      c
0068      c
0069      c
0070      c
0071      c
0072      c
0073      c
0074      c
0075      c
0076      c
0077      c
0078      c
0079      c
0080      c
0081      c
0082      c
0083      c
0084      c
0085      c
0086      c
0087      c
0088      c
0089      c
0090      c
0091      c
0092      c
0093      c
0094      c
0095      c
0096      c
0097      c
0098      c
0099      c
0100      c

```

MUSIC_PLOT_SPECT

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```

0119      character*1 cap
0120      real*4 value_at
0121      c
0122      integer*4 num_ants, select, fskip, count, iskip
0123      real*4 mi_begp, mi_delp
0124      real*8 apars(num_ants,1), wavelen
0125      complex*16 eigvec(num_ants,num_ants)
0126      c
0127      logical replot
0128      integer*4 ch_spc/ 12 /
0129      integer*4 points/ 101 /, x_add, y_add, x_tics/ 5 /, max_tics/ 21 /
0130      integer*4 y_tics, linear( 1 ), status, pts_by_ct( 2 ), plots_add
0131      integer*4 i, ixat, iyat
0132      real*4 phi_start/ - pi4 /, phi_end/ pi4 /, ymin, ymax
0133      real*4 deg_start/ -180. /, deg_end/ 180. /, p, ypeak, yvaly, yat
0134      character*5 numlab
0135      character*10 answer
0136      character*80 msg
0137      c
0138      save points, phi_start, deg_start, phi_end, deg_end, x_tics
0139      c
0140      replot = .true.
0141      do while ( replot )
0142          type *
0143              type 90040, ' Plot MUSIC spectrum from what starting angle (' ,
0144                  1                               deg_start, ') (degrees)? '
0145              accept 90000, answer
0146              read (unit=answer,fmt=*,err=10000) deg_start
0147              phi_start = deg_start * rad_p_deg4
0148          10000      continue
0149          c
0150              type 90040, '                               to what ending angle (' ,
0151                  1                               deg_end, ') (degrees)? '
0152              accept 90000, answer
0153              read (unit=answer,fmt=*,err=10100) deg_end
0154              phi_end = deg_end * rad_p_deg4
0155          10100      continue
0156          c
0157              type 90030, ' Number of points (angles) at which to plot (' ,
0158                  1                               points, ')? '
0159              accept 90000, answer
0160              read (unit=answer,fmt=*,err=10200) points
0161          10200      continue
0162          c
0163              type 90030, ' Number of vertical grid lines to place on the plot (' ,
0164                  1                               x_tics, ')? '
0165              accept 90000, answer
0166              read (unit=answer,fmt=*,err=10300) x_tics
0167          10300      continue
0168          c
0169              if ( phi_start .ge. phi_end ) go to 2000
0170              if ( points .lt. 6 ) go to 2000
0171              if ( x_tics .lt. 0 ) go to 2000
0172              if ( x_tics .gt. max_tics ) go to 2000
0173          c
0174              linear(1) = points
0175              pts_by_ct(1) = points

```

MUSIC_PLOT_SPECT

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```

0176 pts_by_ct(2) = count
0177 call open_va( x_add, linear, 1, sr_type, status, msg )
0178 call exit_va( status, msg )
0179 call open_va( y_add, linear, 1, sr_type, status, msg )
0180 call exit_va( status, msg )
0181 call open_va( plots_add, pts_by_ct, 2, sr_type, status, msg )
0182 call exit_va( status, msg )
0183 c
0184 call music_spectrumx( phi_start, phi_end, points, %val( x_add ) )
0185 do i = 1, count
0186   p = mi_begp + ( fskip + ( i - 1 ) * ( iskip + 1 ) ) * mi_delp
0187   type 90060, 'p =', p, '.'
0188 c
0189   call music_spectrumy( points, apars, num_ants, wavelen,
0190                       eigvec, select, p, %val( x_add ),
0191                       %val( y_add ) )
0192   call r4_col_ins( %val( y_add ), points, %val( plots_add ), i )
0193 c
0194   call vecext( %val( y_add ), points, ymax, ymin )
0195   if ( i .eq. 1 ) then
0196     ypeak = ymax
0197     yvaly = ymin
0198   else
0199     if ( ymax .gt. ypeak ) ypeak = ymax
0200     if ( ymin .lt. yvaly ) yvaly = ymin
0201   end if
0202 end do
0203 call log_scale( ypeak, yvaly, ymax, ymin, y_tics ) .
0204 c
0205 type 90050, ' Ready to plot for phi from ',
0206           1 deg_start, ' to ', deg_end, ' degrees.'
0207 accept 90000, answer
0208 call tek_start
0209 call tek_grid( 200, 150, 922, 691, deg_start, ymin,
0210             deg_end, ymax, x_tics, y_tics, x_tics, y_tics, 0, 1 )
0211 call tek_title( 'Arrival Angle', 'DOA Measure',
0212             1 'DOA SPECTRUM FOR SELECTED p' )
0213 c
0214 call vecscale( %val( x_add ), points, deg_p_rad4 )
0215 do i = 1, count
0216   call r4_col_ext( %val( plots_add ), points, i, %val( y_add ) )
0217   call tek_plot( %val( x_add ), %val( y_add ), points, i )
0218 c
0219   yat = value_at( %val( y_add ), points, .4 )
0220   call tek_point( deg_end, yat, ixat, iyat )
0221   p = mi_begp + ( fskip + ( i - 1 ) * ( iskip + 1 ) ) * mi_delp
0222   write(unit=numlab,fmt=90010) p
0223   call tek_label( numlab( 2 : 5 ), 4, ixat, iyat - 4, ch_spc, 0 )
0224 end do
0225 call tek_vec2( 1, 1, 1, 1, 0 )
0226 c
0227 call tek_finish( .false. )
0228 accept 90000, answer
0229 c
0230 call free_va( x_add, status, msg )
0231 call exit_va( status, msg )
0232 call free_va( y_add, status, msg )

```


MUSIC_PLOT_SPECT

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```
0233      call exit_va( status, msg )
0234      call free_va( plots_add, status, msg )
0235      call exit_va( status, msg )
0236      go to 1000
0237      c
0238      2000      continue
0239              type *, char( 7 )
0240              type *, 'Couldn't plot DOA spectrum using the selected ',
0241                    1      'parameters.'
0242              type *, 'Check for insane phi range or number of points.'
0243      c
0244      1000      continue
0245              type *
0246              type *, 'Enter X (or x) to abort DOA plotting:'
0247              type 90020, '      Do you wish a replot (new phi range) '
0248                    1      // '(y/[n]/x)? '
0249              accept 90000, answer
0250              replot = ( cap( answer( 1 : 1 ) ) .eq. 'Y' )
0251      end do
0252      c
0253      music_plot_spect = ( cap( answer( 1 : 1 ) ) .eq. 'X' )
0254      return
0255      c
0256      90000      format( a120 )
0257      90010      format( f5.2 )
0258      90020      format( $a )
0259      90030      format( $a, i11, a )
0260      90040      format( $a, f7.2, a )
0261      90050      format( $ 2( a, f6.1 ), a )
0262      90060      format( 1x, a, f5.2, a )
0263      c
0264      end
```


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20-Feb-1988 14:33:56 MU

```

0001      real*8 function music_power( eigvec, steer, n, m, p )
0002      c
0003      c-----
0004      c
0005      c          This function returns the MUSIC "power spectrum" sample
0006      c          selected by steering vector steer and "power" exponent p. Division by
0007      c          zero and floating-point underflow and overflow are handled, and the
0008      c          value returned may be inverted without encountering division by zero or
0009      c          floating-point underflow or overflow.
0010      c
0011      c          Parameters:
0012      c          -----
0013      c          eigvec:  n x n COMPLEX*16 matrix whose FIRST m columns are the
0014      c                   "noise" generalized eigenvectors defined in the
0015      c                   MUSIC algorithm.
0016      c          steer:  n-element COMPLEX*16 steering vector (for the chosen
0017      c                   look direction).
0018      c          n and m:  INTEGER*4 index values.
0019      c          p:  REAL*4 power to which to raise the terms used in
0020      c                   accumulating the DOA spectrum sample (p = 2 for
0021      c                   "ordinary" MUSIC).
0022      c
0023      c-----
0024      c
0025      c          implicit none
0026      c          external lib$establish, mpwr_f15
0027      c          external mpwr_dpdtm, mpwr_dpdsun, mpwr_pwrtrm, mpwr_pwrsum
0028      c          include '(fortran_limits)'
0029      c
0030      c          integer*4 lib$establish
0031      c          integer*4 mpwr_dpdtm, mpwr_dpdsun
0032      c          real*8 mpwr_pwrtrm, mpwr_pwrsum
0033      c
0034      c          real*4 pb2lim
0035      c          parameter (pb2lim = 2.0 * minrecip4)
0036      c
0037      c          integer*4 n, m
0038      c          real*4 p
0039      c          complex*16 eigvec(n,n), steer(n)
0040      c
0041      c          integer*4 i, j, old_handler
0042      c          real*8 pb2, pwr_term
0043      c          complex*16 dpd_term, dotprod
0044      c
0045      c          if ( p .ge. pb2lim ) then
0046      c             pb2 = p / 2.0d0
0047      c          else
0048      c             pb2 = 0.0d0
0049      c          end if
0050      c          music_power = 0.0d0
0051      c
0052      c          do i = 1, m
0053      c             dotprod = ( 0.0d0, 0.0d0 )
0054      c
0055      c             do j = 1, n
0056      c                if ( mpwr_dpdtm( eigvec(j,i), steer(j), dpd_term ) )
0057      c                   1
0058      c
0059      c             10000, 10010, 10020

```

```

MUSIC_POWER                                18-Apr-1988 14:24:15  \
                                             20-Feb-1988 14:33:56  P

0086      10000      continue                                ! Underflow.
0087          dpd_term = ( 0.0d0, 0.0d0 )
0088          go to 10010
0089      10020      continue                                ! Overflow.
0090          music_power = minrecip4
0091          return                                        ! Error EXIT!
0092      10010      continue                                ! No error.
0093      c
0094          if ( mpwr_dpdsun( dotprod, dpd_term ) ) 11000, 11010, 10020
0095      11000      continue                                ! Underflow.
0096          dotprod = ( 0.0d0, 0.0d0 )
0097      11010      continue                                ! No error.
0098          end do
0099      c
0100          pwr_term = mpwr_pwrtrm( dotprod, pb2 )
0101          music_power = mpwr_pwrsum( music_power, pwr_term )
0102      end do
0103      c
0104          old_handler = lib$establish( mpwr_f15 )
0105          music_power = 1.0d0 / music_power
0106          call lib$establish( old_handler )
0107      c
0108          if ( music_power .lt. minrecip4 ) then
0109              music_power = minrecip4
0110          else if ( music_power .gt. maxrecip4 ) then
0111              music_power = maxrecip4
0112          end if
0113      c
0114          return
0115          end

```


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```
0001      c
0002      integer*4 function mpwr_dpdtm( eigvec_elem, steer_elem, result )
0003      c
0004      c!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!
0005      c
0006      c          This function places the product
0007      c          dconjg( eigvec_elem ) * steer_elem in the COMPLEX*16 variable
0008      c          result. Arithmetic exceptions are processed by the function
0009      c          MPWR_FL1. The parameters eigvec_elem and steer_elem are COMPLEX*16.
0010      c          The function returns -1 if arithmetic underflow occurred, 0
0011      c          if no arithmetic errors occurred, and +1 if arithmetic overflow or
0012      c          divide by zero occurred.
0013      c
0014      c          implicit none
0015      c          intrinsic dconjg
0016      c          external lib$establish, mpwr_fl1
0017      c
0018      c          integer*4 lib$establish
0019      c
0020      c          complex*16 eigvec_elem, steer_elem, result
0021      c
0022      c          integer*4 old_handler
0023      c
0024      c          old_handler = lib$establish( mpwr_fl1 )
0025      c          result = dconjg( eigvec_elem ) * steer_elem
0026      c          call lib$establish( old_handler )
0027      c
0028      c          mpwr_dpdtm = 0
0029      c          return
0030      c
0031      c          end
```


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```
0001 c
0002 c      real*8 function mpwr_pwrtrm( dotprod, pb2 )
0003 c
0004 c!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!
0005 c
0006 c      This function returns the value of the expression
0007 c      dreal( dotprod * dconjg( dotprod ) ) ** pb2, where dotprod is
0008 c      COMPLEX*16 and pb2 is REAL*8. Arithmetic exceptions are processed
0009 c      by the condition handler MPWR_FL9, below. If floating point
0010 c      underflow occurs, the function return value is corrected to zero. If
0011 c      floating point overflow or divide by zero occur, the maximum
0012 c      invertible REAL*4 value is returned as the function result.
0013 c
0014 c      implicit none
0015 c      intrinsic dreal, dconjg
0016 c      external lib$establish, mpwr_f13
0017 c
0018 c      integer*4 lib$establish
0019 c
0020 c      real*8 pb2
0021 c      complex*16 dotprod
0022 c
0023 c      integer*4 old_handler
0024 c
0025 c      old_handler = lib$establish( mpwr_f13 )
0026 c      mpwr_pwrtrm = dreal( dotprod * dconjg( dotprod ) ) ** pb2
0027 c      call lib$establish( old_handler )
0028 c
0029 c      return
0030 c      end
```

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```

0001 c
0002 integer*4 function mpwr_f13( sigargs, mechargs )
0003 c
0004 c..... Condition handler for MPWR_PWRTRM and MPWR_PWSUM. ....
0005 c
0006 implicit none
0007 external sys$unwind, lib$match_cond, s_to_dr
0008 include 'SYS$LIBRARY:SIGDEF'
0009 include 'SYS$LIBRARY:MTHDEF'
0110 include '(fortran_limits)'
0139 c
0140 integer*4 lib$match_cond
0141 c
0142 integer*4 sigargs( * ), mechargs( * )
0143 c
0144 integer*4 i
0145 c
0146 i = lib$match_cond( sigargs(2),
0147 1 SS$_UNWIND, ! unwind in progress.
0148 2 SS$_FLTUND, ! floating underflow trap.
0149 3 SS$_FLTUNDF, ! floating underflow fault.
0150 4 SS$_FLTOVF, ! floating overflow trap.
0151 5 SS$_FLTDIV, ! floating div. by zero trap.
0152 6 SS$_FLTOVF_F, ! floating overflow fault.
0153 7 SS$_FLTDIV_F ! floating div. by zero fault.
0154 8 )
0155 c
0156 if ( i .eq. 0 ) then ! Unknown condition
0157 mpwr_f13 = SS$_RESIGNAL
0158 else if ( i .eq. 1 ) then ! Unwinding.
0159 else if ( i .lt. 4 ) then ! Underflow.
0160 call s_to_dr( mechargs(4), 0.0 )
0161 call sys$unwind( , )
0162 else ! Overflow.
0163 call s_to_dr( mechargs(4), maxrecip4 )
0164 call sys$unwind( , )
0165 end if
0166 c
0167 return
0168 end

```

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```

0001  c      subroutine s_to_dr( destination, source )
0002
0003  c
0004  c.....      Executes destination = source, REAL*8 destination, .....
0005  c..... REAL*4 source.
0006  c
0007  c      implicit none
0008  c
0009  c      real*4 source
0010  c      real*8 destination
0011  c
0012  c      destination = source
0013  c      return
0014  c
0015  c!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!!
0016  c
0017  c      end
    
```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	8	PIC CON REL LCL SHR EXE RD NOWRT
Total Space Allocated	8	

ENTRY POINTS

Address	Type	Name
0-00000000		S_TO_DR

VARIABLES

Address	Type	Name	Address	Type	Name
AP-00000004@	R*8	DESTINATION	AP-00000008@	R*4	SOURCE

13-Apr-1988 14:24:39 VA:
5-Jan-1988 08:14:47 MU:

```

0001      subroutine music_spectrumx( phi_start, phi_end, phi_num, x_spectrum )
0002      c
0003      c-----
0004      c
0005      c          This subroutine produces abscissa values for the MUSIC "power
0006      c spectrum" for arrival angles between phi_start and phi_end, inclusive.
0007      c
0008      c      Parameters:
0009      c      -----
0010      c          phi_start: REAL*4 minimum angle of arrival for which to
0011      c                      compute the "power" spectrum.
0012      c          phi_end: REAL*4 maximum angle of arrival for which to compute
0013      c                      the "power" spectrum.
0014      c          phi_num: INTEGER*4 number of spectrum samples to compute.
0015      c                      Samples will be computed at intervals of
0016      c                      ( phi_end - phi_start ) / ( phi_num - 1 ).
0017      c          x_spectrum: phi_num-element REAL*4 output vector of abscissa
0018      c                      (phi) values.
0019      c
0020      c-----
0021      c
0022      c      implicit none
0023      c
0024      c      integer*4 phi_num
0025      c      real*4 phi_start, phi_end, x_spectrum(phi_num)
0026      c
0027      c      integer*4 i
0028      c      real*4 phi_del
0029      c
0030      c      phi_del = ( phi_end - phi_start ) / ( phi_num - 1 )
0031      c      do i = 1, phi_num
0032      c          x_spectrum(i) = phi_start + ( i - 1 ) * phi_del
0033      c      end do
0034      c
0035      c      return
0036      c      end

```

18-Apr-1988 14:24:51 Vi
20-Feb-1988 14:35:00 M-

```

0001      subroutine music_spectrumy( phi_num, apars, num_ants, wavelen, eigvec,
0002      1                               num_zeigs, p, x_spectrum, y_spectrum )
0003      c
0004      c-----
0005      c
0006      c          This subroutine produces ordinate values for the MUSIC "power
0007      c          spectrum" for arrival angles in vector x_spectrum, given the applicable
0008      c          generalized eigenvectors and the desired power to which to raise the
0009      c          spectrum point magnitudes.
0010      c
0011      c          Parameters:
0012      c          -----
0013      c          phi_num:  INTEGER*4 number of spectrum samples to compute
0014      c                   (number of elements in x_spectrum and
0015      c                   y_spectrum).
0016      c          apars = matrix whose first column is the vector of ranges (from
0017      c                   the origin) of array antenna elements, and
0018      c                   whose second column is the vector of angles
0019      c                   (from the "x" axis) of array antenna elements
0020      c                   (in RADIANS).  num_ants x 2.  REAL*8.
0021      c                   Note: the position of the ith antenna element is
0022      c                   apars(i,1) * cos( apars(i,2) ) (x position),
0023      c                   apars(i,1) * sin( apars(i,2) ) (y position).
0024      c          num_ants = number of elements in antenna array.  INTEGER*4.
0025      c          wavelen = wavelength of the incident waves.  REAL*8. (Note
0026      c                   that the units of wavelength must be the same
0027      c                   as those in which the array_range(*) are
0028      c                   expressed.)
0029      c          eigvec:  num_ants x num_ants COMPLEX*16 matrix whose FIRST
0030      c                   - num_zeigs columns are the applicable "noise"
0031      c                   generalized eigenvectors.
0032      c          num_zeigs:  INTEGER*4 number of zero generalized eigenvalues
0033      c                   (number of eigenvectors to use in the
0034      c                   computation).
0035      c          p:  REAL*4 power to which to raise the magnitudes of the terms
0036      c                   which make up the reciprocal of a MUSIC DOA
0037      c                   spectrum point (p = 2 for "ordinary" MUSIC).
0038      c          x_spectrum:  phi_num-element REAL*4 vector of abscissa (phi)
0039      c                   values for which to compute the DOA spectrum
0040      c                   values.
0041      c          y_spectrum:  phi_num-element REAL*4 output vector of ordinate
0042      c                   ("power" spectrum sample) values.
0043      c
0044      c-----
0045      c
0046      c          implicit none
0047      c          intrinsic sngl, dble
0048      c          external music_power, signal
0049      c          external open_va, exit_va, free_va
0050      c          include '6000$RT:[CHUCK.RESEARCH.FORTDIR]VATYPES.TXT'
0051      c
0052      c          real*8 music_power
0053      c
0054      c          integer*4 phi_num, num_ants, num_zeigs
0055      c          real*4 p, x_spectrum(phi_num), y_spectrum(phi_num)
0056      c          real*8 apars(num_ants,1), wavelen
0057      c          complex*16 eigvec(num_ants,num_ants)
0058

```

MUSIC_SPECTRUMY

18-Apr-1988 14:24:51

VA:

20-Feb-1988 14:35:00

MU:

```

0069  c
0070      integer*4 i, steerv_add, linear(1), status
0071      character*80 msg
0072  c
0073      linear(1) = num_ants
0074      call open_va( steerv_add, linear, 1, dc_type, status, msg )
0075      call exit_va( status, msg )
0076  c
0077      do i = 1, phi_num
0078          call sigval( apars, num_ants, dble( x_spectrum(i) ), wavelen,
0079                      1, %val( steerv_add ) )
0080          y_spectrum(i) = sngl( music_power( eigvec, %val( steerv_add ),
0081                      1, num_ants, num_zelgs, p ) )
0082      end do
0083  c
0084      call free_va( steerv_add, status, msg )
0085      call exit_va( status, msg )
0086  c
0087      return
0088      end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	270	PIC CON REL LCL SHR EXE RD NOWRT LOI
1 \$PDATA	8	PIC CON REL LCL SHR NOEXE RD NOWRT LOI
2 \$LOCAL	368	PIC CON REL LCL NOSHR NOEXE RD WRT LOI
Total Space Allocated	646	

ENTRY POINTS

Address	Type	Name
0-00000000		MUSIC_SPECTRUMY

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type	Name
**	I*4	I	2-00000004	CHAR	MSG	AP-0000000C@	I*4	NUM_AI
AP-0000001C@	R*4	P	AP-00000004@	I*4	PHI_NUM	2-00000058	I*4	STATU
AP-00000010@	R*8	WAVELEN						

18-Apr-1988 14:25:09 VAX I
 24-Sep-1987 14:59:07 NOISI

```

0001      subroutine noise_combine( mat, n, m, vec, defs, out )
0002      c
0003      c -----
0004      c
0005      c           This subroutine multiplies the n x m REAL*8 matrix mat by the
0006      c           m-element COMPLEX*8 vector vec to produce the n-element COMPLEX*16
0007      c           vector out. The multiplication INCLUDES elements of vec whose
0008      c           corresponding types (high words of first column of n x 2 INTEGER*4
0009      c           matrix defs) are 2 (indicating "burst" noise). These elements are
0010      c           ALSO added to randomly selected elements of out, so the sources are
0011      c           "burst" sources ONLY IF THE APPROPRIATE ENTRIES IN mat ARE ZERO.
0012      c -----
0013      c
0014      c
0015      c           implicit none
0016      c           intrinsic jishft
0017      c           external uni
0018      c
0019      c           real*4 uni
0020      c
0021      c           integer*4 n, m, defs( m, 1 )
0022      c           real*8 mat( n, m )
0023      c           complex*8 vec( m )
0024      c           complex*16 out( n )
0025      c
0026      c           integer*4 i, j
0027      c
0028      c           do i = 1, n
0029      c             out(i) = ( 0.0d0, 0.0d0 )
0030      c             do j = 1, m
0031      c               out(i) = out(i) + mat(i,j) * vec(j)
0032      c             end do
0033      c           end do
0034      c
0035      c           do i = 1, m
0036      c             if ( ( jishft( defs(i,1), -16 ) .eq. 2 )
0037      c               .and. ( vec(i) .ne. ( 0.0, 0.0 ) ) ) then
0038      c               j = 1 + n * uni( 10571 )
0039      c               if ( j .gt. n ) j = n
0040      c               out(j) = out(j) + vec(i)
0041      c             end if
0042      c           end do
0043      c
0044      c           return
0045      c           end

```


18-Apr-1988 14:25:31 VAX
19-Dec-1987 09:16:11 OBS

```

0001      subroutine obsvec( sig_mat, num_ants, num_waves, ndefs, npars,
0002      1                  correl, noises, andefs, anpars, acorrel,
0003      2                  anoises, sig_vec, clean, obs, noise, select )
0004      c
0005      c-----
0006      c
0007      c          This subroutine computes a noisy observation vector from an
0008      c          array of antenna elements in the presence of multiple incident radar
0009      c          waves.
0010      c
0011      c          Parameters:
0012      c          -----
0013      c          sig_mat = num_ants x num_waves COMPLEX*16 matrix of phase
0014      c                   shifts due to antenna position ("signal
0015      c                   matrix").
0016      c          num_ants = number of antenna elements in the array.  INTEGER*4.
0017      c          num_waves = INTEGER*4 number of waves present.
0018      c          ndefs = noises x 2 INTEGER*4 matrix whose first column is the
0019      c                   vector of noise types (see subroutine
0020      c                   READ_NOISE), and whose second column is the
0021      c                   vector of counters for intermittent noise.
0022      c                   This matrix applies to the ESTIMATION NOISE
0023      c                   MODEL.
0024      c          npars = noises x 3 REAL*4 array of noise parameters (see
0025      c                   subroutine READ_NOISE).  This matrix applies to
0026      c                   the ESTIMATION NOISE MODEL.
0027      c          correl = num_ants x noises REAL*8 matrix of noise source
0028      c                   multipliers for each antenna element.  This
0029      c                   matrix applies to the ESTIMATION NOISE MODEL.
0030      c          noises = INTEGER*4 number of noise sources used in the
0031      c                   ESTIMATION NOISE MODEL.
0032      c          andefs = anoises x 2 INTEGER*4 matrix whose first column is the
0033      c                   vector of noise types (see subroutine
0034      c                   READ_NOISE), and whose second column is the
0035      c                   vector of counters for intermittent noise.
0036      c                   This matrix applies to ACTUAL SAMPLE
0037      c                   GENERATION.
0038      c          anpars = anoises x 3 REAL*4 array of noise parameters (see
0039      c                   subroutine READ_NOISE).  This matrix applies to
0040      c                   ACTUAL SAMPLE GENERATION.
0041      c          acorrel = num_ants x anoises REAL*8 matrix of noise source
0042      c                   multipliers for each antenna element.  This
0043      c                   matrix applies to ACTUAL SAMPLE GENERATION.
0044      c          anoises = INTEGER*4 number of noise sources used in ACTUAL
0045      c                   SAMPLE GENERATION.
0046      c          sig_vec = num_waves-element COMPLEX*16 vector of "signal-in-
0047      c                   space" values (complex amplitudes and phases
0048      c                   of signal sources, noise-free, expressed in
0049      c                   "rectangular" coordinates).
0050      c          clean = num_ants-element COMPLEX*16 observation vector, before
0051      c                   corruption by noise.
0052      c          obs = computed observation vector.  num_ants elements.
0053      c                   COMPLEX*16.  Includes noise as determined by
0054      c                   the value of select, below.
0055      c          noise = computed noise vector.  num_ants elements.  COMPLEX*16.
0056      c                   The entries in this vector may be those
0057      c                   actually added to the noise-free observation

```

18-Apr-1988 14:25:31 VA>
19-Dec-1987 09:16:11 DB>

```

0058 c          to produce "obs", or they may be produced by a
0059 c          separate noise model, as determined by
0060 c          "select", below.
0061 c          select = indicator to choose what values of noise are placed in
0062 c          the noise vector. INTEGER*4. Selects
0063 c          "actual" noise values if = 0, selects values
0064 c          from a different ("estimation") noise model if
0065 c          > 0. Note that the parameters "andefs",
0066 c          "anpars", "acorrel", and "anoises" are not
0067 c          referenced unless select > 0.
0068 c
0069 c-----
0070 c
0071 c          implicit none
0072 c          external dc_matvmpy, dc_vecsum, make_noise, dc_veccopy
0073 c          external open_va, free_va, exit_va
0074 c          include '6000$RT:[CHUCK.RESEARCH.FORTDIR]VATYPES.TXT'
0086 c
0087 c          integer*4 num_ants, num_waves, noises, anoises, select
0088 c          integer*4 ndefs( noises, 1 ), andefs( anoises, 1 )
0089 c          real*4 npars( noises, 1 ), anpars( anoises, 1 )
0090 c          real*8 correl( num_ants, noises ), acorrel( num_ants, anoises )
0091 c          complex*16 sig_mat( num_ants, num_waves ), sig_vec( num_waves )
0092 c          complex*16 clean( num_ants ), obs( num_ants ), noise( num_ants )
0093 c
0094 c          integer*4 noisevec_add, va_status, linear( 1 )
0095 c          character*80 va_msg
0096 c
0097 c          linear(1) = num_ants
0098 c          call open_va( noisevec_add, linear, 1, dc_type, va_status, va_msg )
0099 c          call exit_va( va_status, va_msg )
0100 c
0101 c          call dc_matvmpy( sig_mat, num_ants, num_waves, sig_vec, clean )
0102 c          call make_noise( noise, num_ants, ndefs, npars, correl, noises )
0103 c
0104 c          if (select .gt. 0) then
0105 c             call make_noise( %val( noisevec_add ), num_ants,
0106 c                 1, andefs, anpars, acorrel, anoises )
0107 c          else
0108 c             call dc_veccopy( noise, num_ants, %val( noisevec_add ) )
0109 c          end if
0110 c
0111 c          call dc_vecsum( clean, num_ants, %val( noisevec_add ), obs )
0112 c
0113 c          call free_va( noisevec_add, va_status, va_msg )
0114 c          call exit_va( va_status, va_msg )
0115 c
0116 c          return
0117 c          end

```

13-Apr-1988 14:26:03 VA
14-Apr-1988 09:32:39 PI

```

0001      character*1 function
0002      1      pick_music_peak( apars, num_ants, wavelen, eigvec,
0003      2      select, lunit, mi_begp, mi_delp,
0004      3      fskip, num_this, iskip )
0005      c
0006      -----
0007      c
0008      c      This function records peaks in MUSIC "power" spectra according
0009      c      to the user response to the add / delete / replot prompt. During a
0010      c      particular function call, peaks are recorded for num_this values of p,
0011      c      and num_this may be less than or equal to the total number of p values
0012      c      used in the problem.
0013      c
0014      c      Parameters:
0015      c      -----
0016      c      apars = matrix whose first column is the vector of ranges (from
0017      c      the origin) of array antenna elements and whose
0018      c      second column is the vector of angles (from the
0019      c      "x" axis) of array antenna elements (in
0020      c      RADIANS). num_ants x 2. REAL*8.
0021      c      Note: the position of the lth antenna element is
0022      c      apars(l,1) * cos( apars(l,2) ) (x position),
0023      c      apars(l,1) * sin( apars(l,2) ) (y position).
0024      c      num_ants = number of elements in antenna array. INTEGER*4.
0025      c      wavelen = wavelength of the incident waves. REAL*8. (Note
0026      c      that the units of wavelength must be the same
0027      c      as those in which the array_range(*) are
0028      c      expressed.)
0029      c      eigvec: num_ants x num_ants COMPLEX*16 matrix whose FIRST
0030      c      select columns are the applicable "noise"
0031      c      generalized eigenvectors.
0032      c      select: INTEGER*4 number of "noise" eigenvectors (columns of
0033      c      eigvec) to be used in the computation.
0034      c      lunit: INTEGER*4 Logical Unit Number of the file in which to
0035      c      record the estimated angles of arrival (the
0036      c      locations of the detected spectrum peaks).
0037      c      This must be an UNFORMATTED, FIXED record
0038      c      length, 2 quadwords/record, RELATIVE
0039      c      organization, DIRECT access file. The first
0040      c      record holds the number of spectrum peaks
0041      c      detected (updated by this routine).
0042      c      mi_begp: REAL*4 minimum value of the power (p) to which to
0043      c      raise the terms used in accumulating a DOA
0044      c      spectrum sample (p = 2 for "ordinary" MUSIC).
0045      c      mi_delp: REAL*4 minimum increment for p.
0046      c      fskip: INTEGER*4 number of p values to skip before the first
0047      c      p value for which to record the peaks.
0048      c      num_this: INTEGER*4 number of p values for which to record
0049      c      peaks during the current function call.
0050      c      iskip: INTEGER*4 number of p values to skip between values for
0051      c      which to record peaks.
0052      c
0053      -----
0054      c
0055      c      implicit none
0056      c      external cap, spect_peak, readdoa, bulge_doa, crunch_doa, bracket_doa
0057      c      include '(real_constants)'

```

PICK_MUSIC_PEAK

18-Apr-1988 14:26:03

VA

14-Apr-1988 09:32:39

PI

```

0108      c
0109      real*8 spect_peak, readdoa
0110      character*1 cap
0111      c
0112      integer*4 num_ants, select, lunit, fskip, num_this, iskip
0113      real*4 mi_begp, mi_delp
0114      real*8 apars( num_ants, 1 ), wavelen
0115      complex*16 eigvec( num_ants, num_ants )
0116      c
0117      logical first
0118      real*4 p
0119      integer*4 num_picked, p_num, rec_num, i, lleft
0120      real*8 lower/ - pi8 /, upper/ pi8 /, tol/ .01 /, doa
0121      real*8 down/ 3.0d0 /, pwr, langl, uangl, perf
0122      character*8 answer
0123      c
0124      save lower, upper, tol, down
0125      c
0126      pick_music_peak = ' '
0127      do while ( ( pick_music_peak .ne. 'X' )
0128                1 .and. ( pick_music_peak .ne. 'R' )
0129                2 .and. ( pick_music_peak .ne. 'C' ) )
0130          read (unit=lunit,rec=1) num_picked
0131          type *
0132          if ( num_picked .gt. 0 ) then
0133              type 90070, 'Use what # of db down to compute sharpness (',
0134              1 .and. ( pick_music_peak .ne. 'C' )
0135              2 down, 'db)? '
0136          accept 90010, answer
0137          read (unit=answer,fmt=*,err=11000) down
0138          continue
0139      c
0140      type *, 'DOA spectrum peaks currently specified at the following ',
0141      1 .and. ( pick_music_peak .ne. 'C' )
0142      'angles (in degrees):'
0143      type 90060, down
0144      lleft = 19
0145      first = .true.
0146      rec_num = 2
0147      c
0148      do p_num = 1, num_this
0149          p = mi_begp + ( fskip + ( p_num - 1 ) * ( iskip + 1 ) ) * mi_delp
0150          c
0151          do i = 1, num_picked
0152              if ( lleft .le. 0 ) then
0153                  accept 90010, answer
0154                  type 90060, down
0155                  lleft = 20
0156                  first = .true.
0157              else
0158                  lleft = lleft - 1
0159                  first = .false.
0160              end if
0161          c
0162          doa = readdoa( lunit, rec_num )
0163          call bracket_doa( doa, apars, num_ants, wavelen, eigvec,
0164              1 select, p, down, pwr, langl, uangl, perf )
0165          if ( first .or. ( i .eq. 1 ) ) then
0166              type 90020, p, doa * deg_p_rad8, pwr, langl * deg_p_rad8,

```

PICK_MUSIC_PEAK

18-Apr-1988 14:26:03

VAX F

14-Apr-1988 09:32:39

PICK_

```

0165          1          uangi * deg_p_rad8, perf
0166          else
0167              type 90050, doa * deg_p_rad8, pwr, langl * deg_p_rad8,
0168          1          uangi * deg_p_rad8, perf
0169          end if
0170          end do
0171      end do
0172  c
0173      else
0174          type *, 'No DDA spectrum peaks currently specified.'
0175      end if
0176  c
0177      type *
0178      if ( num_picked .eq. 0 ) then
0179          type 90000, ' Add a peak set, replot, continue '
0180          // 'with procedure, or give up (a/r/c/x)? '
0181          accept 90010, pick_music_peak
0182          pick_music_peak = cap( pick_music_peak )
0183          if ( pick_music_peak .eq. 'D' ) pick_music_peak = ' '
0184      else
0185          type *, 'Add or delete a peak set, replot, continue ',
0186          1          'with procedure,'
0187          type 90000, '
0188          // 'or give up (a/d/r/c/x)? '
0189          accept 90010, pick_music_peak
0190          pick_music_peak = cap( pick_music_peak )
0191      end if
0192  c
0193      type *
0194      if ( pick_music_peak .eq. 'A' ) then
0195          type 90040, ' Lower end of peak search interval (degrees) (',
0196          1          lower * deg_p_rad8, ')? '
0197          accept 90010, answer
0198          read (unit=answer,fmt=*,err=10000) lower
0199          lower = lower * rad_p_deg8
0200          10000
0201          continue
0202  c
0203          type 90040, ' Upper end of peak search interval (degrees) (',
0204          1          upper * deg_p_rad8, ')? '
0205          accept 90010, answer
0206          read (unit=answer,fmt=*,err=10010) upper
0207          upper = upper * rad_p_deg8
0208          10010
0209          continue
0209  c
0210          type 90040, ' Locate peak to within how many degrees '
0211          // '( "tolerance" ) (', tol * deg_p_rad8, ')? '
0212          accept 90010, answer
0213          read (unit=answer,fmt=*,err=10020) tol
0214          tol = tol * rad_p_deg8
0215          10020
0216          continue
0215  c
0216          num_picked = num_picked + 1
0217          doa = spect_peak( upper, lower, tol, mi_begp + fskip * mi_delp,
0218          1          apars, num_ants, wavelen, eigvec, select )
0219          rec_num = 2
0220          do while ( rec_num .le. num_picked )
0221              if ( doa .lt. readdoa( lunit, rec_num ) ) then

```

PICK_MUSIC_PEAK

18-Apr-1988 14:26:03
14-Apr-1988 09:32:39VAX
PICK

```

0222             rec_num = rec_num - 1
0223             go to 10040
0224             end if
0225             end do
0226 10040       continue
0227             call bulge_doa( lunit, rec_num - 1, num_this )
0228             write (unit=lunit,rec=rec_num) doa
0229             rec_num = rec_num + num_picked
0230       c
0231             do i = 2, num_this
0232             write (unit=lunit,rec=rec_num)
0233             1       spect_peak( upper, lower, tol, mi_begp +
0234             2         ( fskip + ( i - 1 ) * ( iskip + 1 ) ) * mi_delp,
0235             3         apars, num_ants, wavelen, eigvec, select )
0236             rec_num = rec_num + num_picked
0237             end do
0238       c
0239             else if ( pick_music_peak .eq. 'D' ) then
0240             type 90000, ' Delete which peak (integer) (0 to cancel ' //
0241             1         'deletion)? '
0242             accept 90010, answer
0243             read (unit=answer,fmt=*,err=10030) rec_num
0244             if ( rec_num .gt. 0 ) call crunch_doa( lunit, rec_num, num_this )
0245 10030       continue
0246             end if
0247             end do
0248       c
0249             return
0250       c
0251 90000       format( $a )
0252 90020       format( 1x, 0pf5.2, 4x, 0pf9.4, 1x, 1pg11.4,
0253             1         2( 1x, 0pf9.4 ), 1x, 1pg11.4 )
0254 90010       format( a8 )
0255 90030       format( a, f5.2 )
0256 90040       format( $a, 1pg11.4, a )
0257 90050       format( 10x, 0pf9.4, 1x, 1pg11.4, 2( 1x, 0pf9.4 ), 1x, 1pg11.4 )
0258 90060       format( '           Arrival      Peak      ---', 0pf5.2,
0259             1         'db Down --'
0260             2         / ' p           Angle      Height      Lower      Upper      ',
0261             3         'Sharpness'
0262             4         / ' -----'
0263             5         '-----' )
0264 90070       format( $a, 0pf7.2, a )
0265       c
0266             end

```

18-Apr-1988 14:27:18 VA:
26-Feb-1988 12:02:38 RE:

```

0001      subroutine read_antennae( ant_pars, num_ants, old_ants, def_exp )
0002      c
0003      c-----
0004      c
0005      c          This subroutine sets a group of antenna array element position
0006      c          values, allowing the operator to change the positions if desired.
0007      c          Default values are determined as follows: If old_ants <> num_ants or
0008      c          if def_exp < 0, "standard" defaults are assigned; otherwise default
0009      c          values are assumed to be present in array ant_pars on entry.
0010      c          If necessary, the routine converts operator entries to polar
0011      c          coordinates.
0012      c
0013      c          Parameters:
0014      c          -----
0015      c          ant_pars: num_ants x 2 REAL*8 output matrix whose first column
0016      c          is the vector of antenna array element ranges from the
0017      c          origin, and whose second column is the vector of
0018      c          antenna array element angles from the x axis.
0019      c          num_ants: Number of entries in the ant_range and ant_angle
0020      c          vectors. Also the number of element positions read in.
0021      c          INTEGER*4, input.
0022      c          old_ants: Number of antenna elements in the experiment from
0023      c          which default values (if any) were obtained.
0024      c          INTEGER*4, input.
0025      c          def_exp: Number of the experiment from which default values
0026      c          (if any) were obtained. INTEGER*4, input.
0027      c
0028      c-----
0029      c
0030      implicit none
0031      external ck_polar, cap, nonblank
0032      intrinsic dreal, dimag, dcmplx
0033      include 'real_constants'
0034      c
0035      logical nonblank
0036      integer*4 ck_polar
0037      character*1 cap
0038      c
0039      integer*4 num_ants, old_ants, def_exp
0040      real*8 ant_pars( num_ants, 1 )
0041      c
0042      integer*4 i, j, ant_num, status
0043      real*8 p1, p2
0044      complex*16 position
0045      character*15 instr
0046      c
0047      type *
0048      if ( ( old_ants .eq. num_ants ) .and. ( def_exp .ge. 0 ) ) then
0049      type 90000, ' Use old antennae positions ([Y]/N)? '
0050      else
0051      p2 = pi8t2 / num_ants
0052      do i = 1, num_ants
0053      ant_pars(i,1) = 1.0d0
0054      ant_pars(i,2) = ( i - 1 ) * p2
0055      end do
0056      type 90000, ' Use default antennae positions ([Y]/N)? '
0057      end if
0058      c

```

READ_ANTENNAE

18-Apr-1988 14:27:18

VA

26-Feb-1988 12:02:38

RE

```

0108 c
0109 20000 continue
0110 accept 90010, instr
0111 if ( cap( instr( 1 : 1 ) ) .ne. 'N' ) go to 40000
0112 c
0113 i = 1
0114 30000 continue
0115 type 90020, ' Antenna number (', i, ')? '
0116 accept 90010, instr
0117 if ( nonblank( instr ) ) then
0118 read (unit=instr,fmt=*,err=30000) ant_num
0119 if ( ( ant_num .lt. 1 ) .or. ( ant_num .gt. num_ants ) ) then
0120 type *, char( 7 ), 'Antenna number out of range.'
0121 go to 30000
0122 end if
0123 else
0124 ant_num = i
0125 end if
0126 c
0127 30010 continue
0128 type 90030, ' Range or x (', ant_pars(i,1), ')? '
0129 accept 90010, instr
0130 if ( nonblank( instr ) ) then
0131 read (unit=instr,fmt=*,err=30010) p1
0132 else
0133 p1 = ant_pars(i,1)
0134 end if
0135 c
0136 30020 continue
0137 p2 = ant_pars(i,2) * deg_p_rad8
0138 type 90030, ' Angle (DEGREES) or y (', p2, ')? '
0139 accept 90010, instr
0140 if ( nonblank( instr ) ) then
0141 read (unit=instr,fmt=*,err=30020) p2
0142 end if
0143 c
0144 type 90000, ' Above in polar or rectangular coordinates ([P]/R)? '
0145 accept 90020, instr
0146 if ( cap( instr( 1 : 1 ) ) .eq. 'R' ) then
0147 status = ck_polar( dcmplx( p1, p2 ), position )
0148 if ( status .ne. 0 ) then
0149 type *, char( 7 ), 'Overflow converting to polar coordinates.'
0150 go to 30000
0151 end if
0152 ant_pars(ant_num,1) = dreal( position )
0153 ant_pars(ant_num,2) = dimag( position )
0154 else
0155 ant_pars(ant_num,1) = p1
0156 ant_pars(ant_num,2) = p2 * rad_p_deg8
0157 end if
0158 c
0159 if ( ant_num .eq. i ) i = i + 1
0160 if ( i .le. num_ants ) go to 30000
0161 c
0162 type *
0163 type 90000, ' Entered antenna positions OK ([Y]/N)? '
0164 go to 20000

```


READ_ANTENNAE

18-Apr-1988 14:27:18 VA
26-Feb-1988 12:02:38 RE

```

0165 c
0166 40000 continue
0167 return
0168 c
0169 90000 format( %a )
0170 90010 format( a15 )
0171 90020 format( % a, i5, a )
0172 90030 format( % a, lpg15.7, a )
0173 c
0174 end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 %CODE	1061	PIC CON REL LCL SHR EXE RD NOWRT LC
1 %PDATA	321	PIC CON REL LCL SHR NOEXE RD NOWRT LC
2 %LOCAL	264	PIC CON REL LCL NOSHR NOEXE RD WRT QU
Total Space Allocated	1646	

ENTRY POINTS

Address	Type	Name
0-00000000		READ_ANTENNAE

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type	Name
2-00000030	I*4	ANT_NUM	AP-00000010@	I*4	DEF_EXP	**	I*4	I
**	I*4	J	AP-00000008@	I*4	NUM_ANT	AP-0000000C@	I*4	OLD_A
2-00000018	R*8	P2	2-00000020	C*16	POSITION	**	I*4	STATU

ARRAYS

Address	Type	Name	Bytes	Dimensions
AP-00000004@	R*8	ANT_PARS	**	(* , 1)

LABELS

Address	Label	Address	Label	Address	Label	Address	Label
0-000000CC	20000	0-00000118	30000	0-000001E4	30010	0-0000026C	30020
1-0000012F	90010'	1-00000132	90020'	1-00000138	90030'		

18-Apr-1988 14:27:36 V/
27-Feb-1988 10:21:08 RI

```

0001      subroutine read_lp( def_exp, begp, endp, delp, eps, norm, itlim )
0002      c
0003      c-----
0004      c
0005      c          This subroutine sets parameters for Lp estimation control, and
0006      c          allows the operator to change the values if desired.  if def_exp >= 0,
0007      c          the values of begp, endp, delp, eps, norm, and itlim on entry are
0008      c          assumed to have been obtained from a previous experiment, otherwise
0009      c          are assumed to be "standard" default values.
0010      c
0011      c          Parameters:
0012      c          -----
0013      c          def_exp:  Number of the experiment (if any) from which defaults
0014      c                     were obtained.  INTEGER*4, input.
0015      c          begp:    REAL*4 starting power of residual magnitudes to be
0016      c                     minimized (Lp estimation).
0017      c          endp:    REAL*4 ending power of residual magnitudes to be
0018      c                     minimized (Lp estimation).
0019      c          delp:    REAL*4 increment for power of residual magnitudes to be
0020      c                     minimized.
0021      c          eps:    REAL*8 minimum residual magnitude to use.
0022      c          norm:    REAL*8 value to which to normalize IRLS weights (0.0d0
0023      c                     for non-normalized IRLS).
0024      c          itlim:   INTEGER*4 maximum number of IRLS iterations to perform.
0025      c
0026      c-----
0027      c
0028      c          implicit none
0029      c          external cap, nonblank-
0030      c
0031      c          logical nonblank
0032      c          character*1 cap
0033      c
0034      c          integer*4 def_exp, itlim
0035      c          real*4 begp, endp, delp
0036      c          real*8 eps, norm
0037      c
0038      c          integer*4 count
0039      c          real*8 temp
0040      c          character*15 instr
0041      c
0042      c          type *
0043      c          count = ( endp - begp ) / delp + 1
0044      c          if ( def_exp .ge. 0 ) then
0045      c             type 90040, 'Use old Lp estimation parameters (',
0046      c                 1 count, ' iterations) ([Y]/N)? '
0047      c             else
0048      c                 type 90040, 'Use default Lp estimation parameters (',
0049      c                 1 count, ' iterations) ([Y]/N)? '
0050      c             end if
0051      c
0052      c          20000 continue
0053      c          accept 90010, instr
0054      c          if ( cap( instr( 1 : 1 ) ) .ne. 'N' ) go to 40000
0055      c
0056      c          30000 continue
0057      c          type 90050, ' Lp estimates starting at what p (>= 1) (', begp, ')? '

```

READ_LP

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RE

```

0058      accept 90010, instr
0059      if ( nonblank( instr ) ) then
0060          read (unit=instr,fmt=*,err=30000) temp
0061          if ( ( temp .lt. 1.0 ) .or. ( temp .gt. 3.0 ) ) then
0062              type *, char( 7 ), 'Limits on p are 1 <= p <= 3.'
0063              go to 30000
0064          end if
0065          begp = temp
0066      end if
0067      c
0068      type 90050, '                ending at what p (<= 3) (' , endp, ')? '
0069      accept 90010, instr
0070      if ( nonblank( instr ) ) then
0071          read (unit=instr,fmt=*,err=30000) temp
0072          if ( ( temp .lt. begp ) .or. ( temp .gt. 3.0 ) ) then
0073              type *, char( 7 ), 'Ending p must be >= beginning and <= 3.'
0074              go to 30000
0075          end if
0076          endp = temp
0077      end if
0078      c
0079      if ( endp .gt. begp ) then
0080          type 90050, '                incrementing by what p ( > 0) (' ,
0081      1      delp, ')? '
0082          accept 90010, instr
0083          if ( nonblank( instr ) ) then
0084              read (unit=instr,fmt=*,err=30000) temp
0085              if ( temp .le. 0.0 ) then
0086                  type *, char( 7 ), 'Increment for p must be > 0.'
0087                  go to 30000
0088              end if
0089              delp = temp
0090          end if
0091          else
0092              delp = 3.0
0093          end if
0094      c
0095      30010 continue
0096      type 90020, ' Minimum residual magnitude to use (' , eps, ')? '
0097      accept 90010, instr
0098      if ( nonblank( instr ) ) then
0099          read (unit=instr,fmt=*,err=30010) temp
0100          if ( temp .le. 0.0d0 ) then
0101              type *, char( 7 ), 'Value for epsilon must be greater than 0.'
0102              go to 30010
0103          end if
0104          eps = temp
0105      end if
0106      c
0107      30020 continue
0108      type 90030, ' Normalize weight SQUARE ROOTS to what value (' , norm,
0109      1      ' )? '
0110      type 90000, '                (0 for '
0111      1      // 'unnormalized)? '
0112      accept 90010, instr
0113      if ( nonblank( instr ) ) then
0114          read (unit=instr,fmt=*,err=30020) temp

```

READ_LP

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REAL

```

0115         if ( temp .lt. 0.0d0 ) then
0116             type *, char( 7 ), 'Normalization value must be >= 0.'
0117             go to 30020
0118         end if
0119         norm = temp
0120     end if
0121     c
0122     30030 continue
0123         type 90040, 'Maximum number of IRLS iterations to permit (',
0124         1                                     itlim, ')? '
0125         accept 90010, instr
0126         if ( nonblank( instr ) ) then
0127             read (unit=instr,fmt=*,err=30030) count      ! count scratch.
0128             if ( count .lt. 1 ) then
0129                 type *, char( 7 ), 'Iteration limit must be >= 1.'
0130                 go to 30030
0131             end if
0132             itlim = count
0133         end if
0134     c
0135         type *
0136         count = ( endp - begp ) / delp + 1
0137         type 90040, 'Entered parameters OK (', count,
0138         1                                     ' iterations) (Y/N)? '
0139         go to 20000
0140     c
0141     40000 continue
0142     return
0143     c
0144     90000 format( $a )
0145     90010 format( a80 )
0146     90020 format( $ a, 1pg15.7, a )
0147     90030 format( a, 1pg15.7, a )
0148     90040 format( $, 1x, a, i4, a )
0149     90050 format( $ a, f5.2, a )
0150     c
0151     end

```

18-Apr-1988 14:29:11 V
27-Feb-1988 10:13:28 R

```

0001      subroutine read_misc( def_exp, carrlen, smp_int, tol )
0002      c
0003      c-----
0004      c
0005      c          This subroutine sets miscellaneous parameters required by the
0006      c MUSIC / Lp algorithms, allowing the operator to change the values if
0007      c desired.  If def_exp is >= 0, defaults from a previous experiment are
0008      c assumed to be present in the values carrlen, smp_int, and tol on entry.
0009      c If def_exp is < 0, "standard" defaults are assumed to be present in the
0010      c values.  Note that the value of def_exp serves only to select what
0011      c prompt is used, and that defaults are ASSUMED to be supplied regardless
0012      c of the value of def_exp.
0013      c
0014      c          Parameters:
0015      c          -----
0016      c          def_exp:  Number of the experiment (if any) from which defaults
0017      c                    were obtained.  INTEGER*4, input.
0018      c          carrlen: REAL*8 carrier wavelength of incident waves.  Units
0019      c                    MUST be the same as those in which antenna
0020      c                    ranges are measured.
0021      c          smp_int: REAL*8 sample interval, in carrier wavelengths.
0022      c          tol:    REAL*8 "tolerance"; value to be taken for zero.
0023      c-----
0024      c
0025      c          implicit none
0026      c          external cap, nonblank
0027      c
0028      c          logical nonblank
0029      c          character*1 cap
0030      c
0031      c          integer*4 def_exp
0032      c          real*8 carrlen, smp_int, tol
0033      c
0034      c          character*1 nsel
0035      c          character*90 instr
0036      c
0037      c          type *
0038      c          if ( def_exp .ge. 0 ) then
0039      c              type 90000, ' Use old wavelength, sample interval, and tolerance '
0040      c              1 // '([Y]/N)? '
0041      c          else
0042      c              type 90000, ' Use default wavelength, sample interval, and '
0043      c              1 // 'tolerance ([Y]/N)? '
0044      c          end if
0045      c
0046      c          20000 continue
0047      c          accept 90010, instr
0048      c          if ( cap( instr( 1 : 1 ) ) .ne. 'N' ) go to 40000
0049      c
0050      c          30000 continue
0051      c          type 90020, ' Carrier wavelength (', carrlen, ')? '
0052      c          accept 90010, instr
0053      c          if ( nonblank( instr ) ) read (unit=instr,fmt=*,err=30000) carrlen
0054      c
0055      c          30010 continue
0056      c          type 90020, ' Sample interval (', smp_int,
0057

```

```

READ_MMISC                                18-Apr-1988 14:28:11  VAX F
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0058          1                          ' ) (Carrier Wavelengths)? '
0059          accept 90010, instr
0060          if ( nonblank( instr ) ) read (unit=instr,fmt=*,err=30010) smp_int
0061          c
0062          30020  continue
0063          type 90020, ' Tolerance (value to take for zero) (', tol, ')? '
0064          accept 90010, instr
0065          if ( nonblank( instr ) ) read (unit=instr,fmt=*,err=30020) tol
0066          c
0067          type *
0068          type 90000, ' Entered parameters OK (Y/N)? '
0069          go to 20000
0070          c
0071          40000  continue
0072          return
0073          c
0074          90000  format( $a )
0075          90010  format( a80 )
0076          90020  format( $ a, 1pg15.7, a )
0077          c
0078          end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	609	PIC CON REL LCL SHR EXE RD NOWRT LONG
1 \$PDATA	282	PIC CON REL LCL SHR NOEXE RD NOWRT LONG
2 \$LOCAL	188	PIC CON REL LCL NOSHR NOEXE RD WRT LONG
Total Space Allocated	1079	

ENTRY POINTS

Address	Type	Name
0-00000000		READ_MMISC

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type	Name
AP-00000008	R*8	CARRLEN	AP-00000004	I*4	DEF_EXP	2-00000000	CHAR	INSTR
AP-0000000C	R*8	SMP_INT	AP-00000010	R*8	TOL			

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```

0001      subroutine read_noise( def_exp, ants, old_ants, noises, old_noises,
0002      1                                defs, pars, correl, prompt_sel, old_which )
0003      c
0004      c-----
0005      c
0006      c      This subroutine sets a group of noise characteristic values,
0007      c      permitting the operator to change the characteristics, if desired.
0008      c      The value prompt_sel determines whether "actual" (prompt_sel < 0) or
0009      c      "model" (prompt_sel >= 0) noise generators are being defined.
0010      c      Defaults are determined for "actual" noise generators as
0011      c      follows:  If def_exp > 0, noises = old_noises, and old_which > 0, then
0012      c      default values are assumed to be present in the arrays defs and pars.
0013      c      If, in addition, ants = old_ants, then default values are assumed to be
0014      c      present in array correl on entry as well.  If default values are not
0015      c      assumed to be present in an array, then "standard" defaults are
0016      c      assigned.  For "model" noise generators, defaults are established in
0017      c      the same way except that old_which is ignored (since all experiments
0018      c      use "model" noise generators).
0019      c
0020      c      Parameters:
0021      c      -----
0022      c      def_exp:  INTEGER*4 number of the experiment from which default
0023      c              values (if any) were obtained.
0024      c      ants:    INTEGER*4 number of elements in the antenna array.
0025      c      old_ants: INTEGER*4 number of antenna array elements in the
0026      c              experiment from which default values
0027      c              (if any) were obtained.
0028      c      noises:  INTEGER*4 number of noise generators to be used.
0029      c      old_noises: INTEGER*4 number of noise generators used in the
0030      c              experiment from which default values
0031      c              (if any) were obtained.
0032      c      defs:    noises x 2 INTEGER*4 matrix whose first column is the
0033      c              vector of generator noise types (see
0034      c              below).
0035      c      pars:    noises x 3 REAL*4 array of noise parameters for each
0036      c              antenna (mean, std. dev., ave. rep.
0037      c              rate).
0038      c      correl:  ants x noises REAL*8 matrix of noise source
0039      c              multipliers for each antenna element
0040      c              (noise correlation matrix).
0041      c      prompt_sel: INTEGER*4 value indicating whether the noise
0042      c              sources being defined are those for
0043      c              actual sample generation (prompt_sel
0044      c              < 0), those for the estimator noise
0045      c              model (prompt_sel > 0), or both (i.e.
0046      c              estimators use actual noise)
0047      c              (prompt_sel = 0).
0048      c      old_which: INTEGER*4 value indicating whether actual
0049      c              (old_which = 0) or model (old_which >
0050      c              0) noise generators were used in the
0051      c              estimation routines of the experiment
0052      c              from which default values (if any) were
0053      c              obtained.
0054      c
0055      c      Noise types:
0056      c      -----
0057      c      type = hhll (hex):  hh = high word (16 bits), ll = low word.

```

19-Apr-1988 14:28:25 VA
27-Feb-1988 10:00:08 RE

```

0058 c          ll = 1 => fixed noise (constant amplitude and phase);
0059 c                    pars( i, 1 ) = real part of noise.
0060 c                    pars( i, 2 ) = imaginary part of noise.
0061 c          ll = 2 => uniform noise; pars( i, 1 ) = mean (real & imag.).
0062 c                    pars( i, 2 ) = width (real & imag.).
0063 c          ll = 3 => gaussian noise; pars( i, 1 ) as for uniform.
0064 c                    pars( i, 2 ) = std. dev. (real & i.).
0065 c          ll = others => not used.
0066 c          hh = 0 => noise type determined by ll.
0067 c          hh = 1 => impulsive noise, type determined by ll;
0068 c                    pars( i, 3 ) = average repetit. rate.
0069 c          hh = 2 => burst noise, type determined by ll;
0070 c                    (applied to randomly selected anten.),
0071 c                    (all multipliers = 0.0d0),
0072 c                    pars( i, 3 ) = average repetit. rate.
0073 c
0074 c -----
0075 c
0076 c          implicit none
0077 c          intrinsic jshft, jrand, jior, dcmplx, dble
0078 c          intrinsic sngl, dreal, dimag, alog10, jmod
0079 c          external cap, nonblank, pr_ntype, rect, ck_polar
0080 c          include '(fortran_limits)'
0081 c          include '(real_constants)'
0082 c
0083 c          logical nonblank
0084 c          integer*4 ck_polar
0085 c          complex*16 rect
0086 c          character*1 cap
0087 c
0088 c          integer*4 def_exp, ants, old_ants, noises, old_noises
0089 c          integer*4 defs( noises, 1 ), prompt_sel, old_which
0090 c          real*4 pars( noises, 1 )
0091 c          real*8 correl( ants, noises )
0092 c
0093 c          logical have_one
0094 c          integer*4 i, j, noise_num, it1, it2, ant_fnoise
0095 c          real*4 t1, t2
0096 c          complex*16 z1
0097 c          character*15 instr
0098 c
0099 c          type *
0100 c          if ( prompt_sel .lt. 0 ) then
0101 c             type *, '          Noise sources to be used in ',
0102 c             1 'ACTUAL SAMPLE GENERATION'
0103 c          else if ( prompt_sel .eq. 0 ) then
0104 c             type *, ' Noise sources for BOTH actual sample ',
0105 c             1 'generation AND estimation noise model'
0106 c          else
0107 c             type *, '          Noise sources to be used in the ',
0108 c             1 'ESTIMATION NOISE MODEL'
0109 c          end if
0110 c          type *
0111 c
0112 c          if ( ( def_exp .lt. 0 )
0113 c             1 .or. ( noises .ne. old_noises )
0114 c             2 .or. ( ( prompt_sel .lt. 0 ) .and. ( old_which .le. 0 ) )

```


READ_NOISE

18-Apr-1988 14:29:25
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```

0193      3      ) then
0194      do i = 1, noises
0195      defs(i,1) = 3
0196      pars(i,1) = 0.0
0197      pars(i,2) = 0.03162278
0198      pars(i,3) = 0.0
0199      end do
0200      go to 10000
0201      else
0202      if ( ants .eq. old_ants ) then
0203      type 90000, ' Use old noise generators ([Y]/N)? '
0204      else
0205      10000      continue
0206      do i = 1, ants
0207      do j = 1, noises
0208      if ( i .eq. j ) then
0209      correl(i,j) = 1.0d0
0210      else
0211      correl(i,j) = 0.0d0
0212      end if
0213      end do
0214      end do
0215      type 90000, ' Use default noise generators ([Y]/N)? '
0216      end if
0217      end if
0218      c
0219      20000      continue
0220      accept 90010, instr
0221      if ( cap( instr( 1 :_1 ) ) .ne. 'N' ) go to 40000
0222      c
0223      i = 1
0224      30000      continue
0225      type 90020, ' Noise source number (', i, ')? '
0226      accept 90010, instr
0227      if ( nonblank( instr ) ) then
0228      read (unit=instr,fmt=*,err=30000) noise_num
0229      if ( ( noise_num .lt. 1 ) .or. ( noise_num .gt. noises ) ) then
0230      type *, char( 7 ), 'Source number out of range.'
0231      go to 30000
0232      end if
0233      else
0234      noise_num = i
0235      end if
0236      c
0237      30010      continue
0238      it1 = jshft( defs(i,1), -16 )
0239      if ( it1 .eq. 0 ) then
0240      type 90000, ' Continuous, Intermittent, or Burst noise '
0241      1      // '(Continuous) (C, I, or B)? '
0242      else if ( it1 .eq. 1 ) then
0243      type 90000, ' Continuous, Intermittent, or Burst noise '
0244      1      // '(Intermittent) (C, I, or B)? '
0245      else
0246      type 90000, ' Continuous, Intermittent, or Burst noise '
0247      1      // '(Burst) (C, I, or B)? '
0248      end if
0249      accept 90010, instr

```

READ_NOISE

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RE/

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0250      instr( 1 : 1 ) = cap( instr( 1 : 1 ) )
0251      if ( nonblank( instr ) ) then
0252          if ( instr( 1 : 1 ) .eq. 'C' ) then
0253              it1 = 0
0254          else if ( instr( 1 : 1 ) .eq. 'I' ) then
0255              it1 = 1
0256          else if ( instr( 1 : 1 ) .eq. 'B' ) then
0257              it1 = 2
0258          else
0259              type *, char( 7 ), 'Please enter C<CR>, c<CR>, I<CR>, ',
0260          1          'i<CR>, B<CR>, b<CR>, or <CR>.'
0261              go to 30010
0262          end if
0263      end if
0264      c
0265      30030 continue
0266          it2 = jland( defs(i,1), 65535 )
0267          call pr_notype( it2 )
0268          accept 90010, instr
0269          if ( nonblank( instr ) ) then
0270              it2 = index( 'FUG', cap( instr( 1 : 1 ) ) )
0271          end if
0272          if ( it2 .eq. 0 ) go to 30030
0273          defs(noise_num,1) = jior( jishft( it1, 16 ), it2 )
0274      c
0275          if ( it2 .eq. 1 ) then
0276              if ( ck_polar( dcmplx( dble( pars(i,1) ), dble( pars(i,2) ) ), z1 )
0277          1          .ne. 0 ) z1 = ( 1.0d0, 0.0d0 )
0278              t1 = sngl( drealm( z1 ) )
0279              if ( t1 .lt. minrecip4 ) t1 = 1.0
0280              t2 = sngl( dimag( z1 ) ) * deg_p_rad4
0281          else
0282              t1 = pars(i,1)
0283              if ( pars(i,2) .ge. minrecip4 ) then
0284                  t2 = pars(i,2)
0285              else
0286                  t2 = 1.0
0287              end if
0288          end if
0289          30040 continue
0290              if ( it2 .eq. 1 ) then
0291                  type 90030, ' Noise magnitude (', 20.0 * alog10( t1 ), ') (db)? '
0292              else
0293                  type 90030, ' Distribution mean (', t1, ')? '
0294              end if
0295              accept 90010, instr
0296              if ( nonblank( instr ) ) then
0297                  read (unit=instr,fmt=*,err=30040) pars(noise_num,1)
0298                  if ( it2 .eq. 1 ) pars(noise_num,1) =
0299          1          10.0 ** ( pars(noise_num,1) / 20.0 )
0300              else
0301                  pars(noise_num,1) = t1
0302              end if
0303      c
0304      30050 continue
0305          if ( it2 .eq. 1 ) then
0306              type 90030, ' Noise phase (', t2, ') (DEGREES)? '

```

READ_NOISE

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```

0307         else
0308             type 90030, ' Noise power (', 20.0 * alog10( t2 ), ') (db)? '
0309         end if
0310         accept 90010, instr
0311         if ( nonblank( instr ) ) then
0312             read (unit=instr,fmt=*,err=30050) pars(noise_num,2)
0313             if ( it2 .ne. 1 ) pars(noise_num,2) =
0314             1             10.0 ** ( pars(noise_num,2) / 20.0 )
0315         else
0316             pars(noise_num,2) = t2
0317         end if
0318         if ( it2 .eq. 1 ) then
0319             z1 = rect( dcmplx( dble( pars(noise_num,1) ),
0320             1             rad_p_deg8 * pars(noise_num,2) ) )
0321             pars(noise_num,1) = sngl( dreal( z1 ) )
0322             pars(noise_num,2) = sngl( dimag( z1 ) )
0323         end if
0324     c
0325 30060 continue
0326     if ( it1 .ne. 0 ) then
0327         type 90030,
0328         1         ' Average repetition interval( ', pars(i,3), ') (samples)? '
0329         accept 90010, instr
0330         if ( nonblank( instr ) ) then
0331             read (unit=instr,fmt=*,err=30060) pars(noise_num,3)
0332         else
0333             pars(noise_num,3) = pars(i,3)
0334         end if
0335     end if
0336     c
0337     if ( it1 .ne. 2 ) then
0338         have_one = .false.
0339 30070 continue
0340         if ( have_one ) then
0341             type 90000, ' Apply this source at which antenna (<CR> for '
0342             1             // 'next source)? '
0343         else
0344             ant_fnoise = jmod( noise_num, ants )
0345             if ( ant_fnoise .eq. 0 ) ant_fnoise = ants
0346             type 90020, ' Apply this source at which antenna (', ant_fnoise,
0347             1             ')? '
0348         end if
0349         accept 90010, instr
0350         if ( nonblank( instr ) ) then
0351             read (unit=instr,fmt=*,err=30075) j
0352             if ( ( j .lt. 1 ) .or. ( j .gt. ants ) ) go to 30075
0353             if ( have_one ) then
0354                 type 90030, '
0355             1             Multiplier (',
0356                 correl(j,i), ')? '
0357             else
0358                 type 90030, '
0359                 Multiplier( ', correl(j,i), ')? '
0360             end if
0361             accept 90010, instr
0362             if ( nonblank( instr ) ) then
0363                 read (unit=instr,fmt=*,err=30075) correl(j,noise_num)
0364             else
0365                 correl(j,noise_num) = correl(j,i)

```

READ_NOISE

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REAL

```

0364         end if
0365         have_one = .true.
0366         go to 30070
0367 30075      continue
0368           type *, char( 7 ), 'Error on input. Please re-enter ',
0369           'from start.'
0370           go to 30070
0371         else if ( .not. have_one ) then
0372           j = ant_fnoise
0373           type 90030, ' Multiplier(', correl(j,i), ')? '
0374           accept 90010, instr
0375           if ( nonblank( instr ) ) then
0376             read (unit=instr,fmt=*,err=30075) correl(j,noise_num)
0377           else
0378             correl(j,noise_num) = correl(j,i)
0379           end if
0380           have_one = .true.
0381           go to 30070
0382         end if
0383       else
0384         do j = 1, ants
0385           correl(j,noise_num) = 0.0d0
0386         end do
0387       end if
0388     c
0389     if ( noise_num .eq. i ) i = i + 1
0390     if ( i .le. noises ) go to 30000
0391     c
0392     type *
0393     type 90000, ' Entered noise parameters OK ([Y]/N)? '
0394     go to 20000
0395     c
0396 40000    continue
0397     return
0398     c
0399 90000    format( $ a )
0400 90010    format( a15 )
0401 90020    format( $ a, i5, a )
0402 90030    format( $ a, 1pg15.7, a )
0403     c
0404     end

```

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```

0001      c
0002      c      subroutine pr_ntype( ntype )
0003      c
0004      c*****
0005      c
0006      c      This subroutine prompts for noise distribution type. The type
0007      c      of noise distribution included in the prompt (the default type) is
0008      c      selected by the INTEGER*4 value ntype (1 = fixed (constant magnitude
0009      c      and phase), 2 = uniform, 3 = gaussian)
0010      c
0011      c*****
0012      c
0013      c      integer*4 ntype
0014      c
0015      c      character*21 head/' Noise distribution ('
0016      c      character*17 tail/')(F, U, or G)? '/
0017      c
0018      c      if ( ntype .eq. 1 ) type 90000, head // 'Fixed' // tail
0019      c      if ( ntype .eq. 2 ) type 90000, head // 'Uniform' // tail
0020      c      if ( ntype .eq. 3 ) type 90000, head // 'Gaussian' // tail
0021      c
0022      c      return
0023      c
0024      90000      format ( $a )
0025      c
0026      c      end
    
```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	226	PIC CON REL LCL SHR EXE RD NOWRT
1 \$PDATA	15	PIC CON REL LCL SHR NOEXE RD NOWRT
2 \$LOCAL	48	PIC CON REL LCL NOSHR NOEXE RD WRT
Total Space Allocated		289

ENTRY POINTS

Address	Type	Name
0-00000000		PR_NTTYPE

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type	Name
2-00000000	CHAR	HEAD	AP-00000004@	1*4	NTTYPE	2-00000015	CHAR	TAI

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```

0001      subroutine read_waves( num_waves, old_waves, def_exp,
0002      1                          wave_pars, mtypes )
0003      c
0004      c-----
0005      c
0006      c      This subroutine sets a group of wave characteristic values,
0007      c      allowing the operator to change the values if desired. Default values
0008      c      are determined as follows: If old_waves <> num_waves or def_exp < 0,
0009      c      "standard" defaults are assigned, otherwise default values are assumed
0010      c      to be present in arrays wave_pars and mtypes on entry.
0011      c
0012      c      Parameters:
0013      c      -----
0014      c      num_waves: Number of rows in the output matrices. Also the
0015      c                  number of wave parameter sets read in.
0016      c                  INTEGER*4, Input.
0017      c      old_waves: Number of incident waves in the experiment from
0018      c                  which default values (if any) were obtained.
0019      c                  INTEGER*4, Input.
0020      c      def_exp: Number of the experiment from which default values
0021      c                  (if any) were obtained. INTEGER*4, Input.
0022      c      wave_pars: num_waves x 9 REAL*8 matrix whose first 8 columns
0023      c                  are as follows:
0024      c                  1) vector of strengths of the waves, expressed
0025      c                      in db (referred to arbitrary unit
0026      c                      amplitude).
0027      c                  2) vector of starting carrier phases of the
0028      c                      arriving waves, RADIANS.
0029      c                  3) vector of wave angles of arrival, from the x
0030      c                      axis, RADIANS.
0031      c                  4) vector of modulation cycle lengths, in
0032      c                      number of carrier cycles per modulation
0033      c                      cycle.
0034      c                  5) vector of modulation duty cycles (expressed
0035      c                      as fractions of modulation cycle
0036      c                      length, 0.0 < mod_duty(i) <= 1.0)
0037      c                      (applicable only to ramp and
0038      c                      rectangular modulation).
0039      c                  6) vector of "starting" modulation phases, in
0040      c                      number of carrier cycles.
0041      c                  7) vector of amplitude modulation ratios
0042      c                      (maximum / minimum).
0043      c                  8) vector of phase modulation total phase
0044      c                      shifts (in RADIANS).
0045      c      mtypes: num_waves x 2 INTEGER*4 matrix whose first column is
0046      c                  the vector of amplitude modulation types (see
0047      c                  below), and whose second column is the vector
0048      c                  of phase modulation types (see below).
0049      c
0050      c      Modulation types: 1 = sine, 2 = triangle, 3 = increasing ramp,
0051      c      4 = decreasing ramp, 5 = rectangular.
0052      c
0053      c-----
0054      c
0055      c      implicit none
0056      c      intrinsic index
0057      c      external cap, nonblank, pr_mtype

```

```

)
)
) READ_WAVES
)
) 0058      include '(real_constants)'
) 0109      c
) 0110      logical nonblank
) 0111      character*1 cap
) 0112      c
) 0113      integer*4 num_waves, old_waves, def_exp
) 0114      integer*4 mtypes( num_waves, 1 )
) 0115      real*8 wave_pars( num_waves, 1 )
) 0116      c
) 0117      integer*4 i, j, wave_num, lt1
) 0118      real*8 t1, t2, t3
) 0119      character*15 instr
) 0120      c
) 0121      type *
) 0122      if ( ( old_waves .eq. num_waves ) .and. ( def_exp .ge. 0 ) ) then
) 0123          type 90000, ' Use old wave characteristics (Y/N)? '
) 0124      else
) 0125          t1 = 1.0d0 / num_waves
) 0126          t2 = 30.0d0 * t1
) 0127          t3 = pi8t2 * t1
) 0128          do i = 1, num_waves
) 0129              wave_pars(i,1) = i * t2
) 0130              wave_pars(i,2) = 0.0d0
) 0131              wave_pars(i,3) = ( i - 1 ) * t3
) 0132              wave_pars(i,4) = 30.0d0
) 0133              wave_pars(i,5) = t1
) 0134              wave_pars(i,6) = ( i - 1 ) * t2
) 0135              wave_pars(i,7) = 100.0d0
) 0136              wave_pars(i,8) = 0.0d0
) 0137              mtypes(i,1) = 5
) 0138              mtypes(i,2) = 5
) 0139          end do
) 0140          type 90000, ' Use default wave characteristics (Y/N)? '
) 0141      end if
) 0142      c
) 0143      20000 continue
) 0144      accept 90010, instr
) 0145      if ( cap( instr( 1 : 1 ) ) .ne. 'N' ) go to 40000
) 0146      c
) 0147      i = 1
) 0148      30000 continue
) 0149          type 90020, ' Wave number (', i, ')? '
) 0150          accept 90010, instr
) 0151          if ( nonblank( instr ) ) then
) 0152              read (unit=instr,fmt=*,err=30000) wave_num
) 0153              if ( ( wave_num .lt. 1 ) .or. ( wave_num .gt. num_waves ) ) then
) 0154                  type *, char( 7 ), 'Wave number out of range.'
) 0155                  go to 30000
) 0156              end if
) 0157          else
) 0158              wave_num = i
) 0159          end if
) 0160      c
) 0161      30010 continue
) 0162          type 90030, ' Wave strength (', wave_pars(i,1), ') (db)? '
) 0163          accept 90010, instr
) 0164          if ( nonblank( instr ) ) then

```

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READ WAVES

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```

0165         read (unit=instr,fmt=*,err=30010) wave_pars(wave_num,1)
0166     else
0167         wave_pars(wave_num,1) = wave_pars(i,1)
0168     end if
0169     c
0170 30015 continue
0171     type 90030, ' Carrier start phase (', wave_pars(i,2) * deg_p_rad8,
0172     1                                     ') (DEGREES)? '
0173     accept 90010, instr
0174     if ( nonblank( instr ) ) then
0175         read (unit=instr,fmt=*,err=30015) wave_pars(wave_num,2)
0176         wave_pars(wave_num,2) = wave_pars(wave_num,2) * rad_p_deg8
0177     else
0178         wave_pars(wave_num,2) = wave_pars(i,2)
0179     end if
0180     c
0181 30020 continue
0182     type 90030, ' Angle of arrival (', wave_pars(i,3) * deg_p_rad8,
0183     1                                     ') (DEGREES)? '
0184     accept 90010, instr
0185     if ( nonblank( instr ) ) then
0186         read (unit=instr,fmt=*,err=30020) wave_pars(wave_num,3)
0187         wave_pars(wave_num,3) = wave_pars(wave_num,3) * rad_p_deg8
0188     else
0189         wave_pars(wave_num,3) = wave_pars(i,3)
0190     end if
0191     c
0192     t1 = wave_pars(i,6) / wave_pars(i,4)
0193 30030 continue
0194     type 90030, ' Modulation length (', wave_pars(i,4),
0195     1                                     ') (Carrier Cycles) (>=1)? '
0196     accept 90010, instr
0197     if ( nonblank( instr ) ) then
0198         read (unit=instr,fmt=*,err=30030) t2
0199     else
0200         t2 = wave_pars(i,4)
0201     end if
0202     if ( t2 .lt. 1.0d0 ) go to 30030
0203     wave_pars(wave_num,4) = t2
0204     c
0205 30040 continue
0206     type 90030, ' Modulation starting phase (', t1 * 360.0d0,
0207     1                                     ') (DEGREES)? '
0208     accept 90010, instr
0209     if ( nonblank( instr ) ) then
0210         read (unit=instr,fmt=*,err=30040) wave_pars(wave_num,6)
0211         wave_pars(wave_num,6) =
0212     1     ( wave_pars(wave_num,6) / 360.0d0 ) * wave_pars(wave_num,4)
0213     else
0214         wave_pars(wave_num,6) = t1 * wave_pars(wave_num,4)
0215     end if
0216     c
0217 30050 continue
0218     type 90030, ' Modulation duty cycle (', wave_pars(i,5),
0219     1                                     ') (>0, <=1)? '
0220     accept 90010, instr
0221     if ( nonblank( instr ) ) then

```


READ_WAVES

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```

0222         read (unit=instr,fmt=*,err=30050) t1
0223     else
0224         t1 = wave_pars(i,5)
0225     end if
0226     if ( ( t1 .le. 0.0d0 ) .or. ( t1 .gt. 1.0d0 ) ) go to 30050
0227     wave_pars(wave_num,5) = t1
0228 c
0229 30060 continue
0230     call pr_mtype( 'Amplitude', mtypes(i,1) )
0231     accept 90010, instr
0232     if ( nonblank( instr ) ) then
0233         it1 = index( 'STIDR', cap( instr( 1 : 1 ) ) )
0234     else
0235         it1 = mtypes(i,1)
0236     end if
0237     if ( it1 .eq. 0 ) go to 30060
0238     mtypes(wave_num,1) = it1
0239 c
0240 30070 continue
0241     type 90030, ' Amplitude modulation ratio (', wave_pars(i,7),
0242     1         ' ) (max / min)? '
0243     accept 90010, instr
0244     if ( nonblank( instr ) ) then
0245         read (unit=instr,fmt=*,err=30070) wave_pars(wave_num,7)
0246     else
0247         wave_pars(wave_num,7) = wave_pars(i,7)
0248     end if
0249 c
0250 30080 continue
0251     call pr_mtype( 'Phase', mtypes(i,2) )
0252     accept 90010,-instr
0253     if ( nonblank( instr ) ) then
0254         it1 = index( 'STIDR', cap( instr( 1 : 1 ) ) )
0255     else
0256         it1 = mtypes(i,2)
0257     end if
0258     if ( it1 .eq. 0 ) go to 30080
0259     mtypes(wave_num,2) = it1
0260 c
0261 30090 continue
0262     type 90030, ' Phase modulation total shift (',
0263     1         wave_pars(i,8) * deg_p_rad8, ' ) (DEGREES)? '
0264     accept 90010, instr
0265     if ( nonblank( instr ) ) then
0266         read (unit=instr,fmt=*,err=30090) wave_pars(wave_num,8)
0267         wave_pars(wave_num,8) = wave_pars(wave_num,8) * rad_p_deg8
0268     else
0269         wave_pars(wave_num,8) = wave_pars(i,8)
0270     end if
0271 c
0272     if ( wave_num .eq. i ) i = i + 1
0273     if ( i .le. num_waves ) go to 30000
0274 c
0275     type *
0276     type 90000, ' Entered wave characteristics OK (Y/N)? '
0277     go to 20000
0278 c

```

READ_WAVES

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```
0279      c              Sort entered wave parameters into ascending DOA order.
0280      c
0281      40000  continue
0282      do i = 1, num_waves - 1
0283          wave_num = i
0284          do j = i + 1, num_waves
0285              if ( wave_pars(j,3) .lt. wave_pars(wave_num,3) ) wave_num = j
0286          end do
0287          if ( wave_num .ne. i ) then
0288              do j = 1, 8
0289                  t1 = wave_pars(i,j)
0290                  wave_pars(i,j) = wave_pars(wave_num,j)
0291                  wave_pars(wave_num,j) = t1
0292              end do
0293              do j = 1, 2
0294                  it1 = mtypes(i,j)
0295                  mtypes(i,j) = mtypes(wave_num,j)
0296                  mtypes(wave_num,j) = it1
0297              end do
0298          end if
0299      end do
0300      c
0301      return
0302      c
0303      90000  format( $ a )
0304      90010  format( a15 )
0305      90020  format( $ a, 15, a )
0306      90030  format( $ a, 1pg15.7, a )
0307      c
0308      end
```

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26-Feb-1988 12:03:36

```

0001 c
0002 c      subroutine pr_mtype( kind, ptype )
0003 c
0004 c*****
0005 c
0006 c      This subroutine prompts for modulation type.  The CHARACTER*
0007 c      string kind is prepended to the prompt, and the type of modulation
0008 c      included in the prompt (the default type) is selected by the INTEGER
0009 c      value ptype (1 = sine, 2 = triangular, 3 = increasing ramp,
0010 c      4 = decreasing ramp, 5 = rectangular).
0011 c
0012 c*****
0013 c
0014 c      integer*4 ptype
0015 c      character*(*) kind
0016 c
0017 c      character*18 head// modulation type (//
0018 c      character*23 tail//) (S, T, I, D, or R)? //
0019 c
0020 c      if ( ptype .eq. 1 ) type 90000,
0021 c      1 // kind // head // 'Sinusoidal' // tail
0022 c      if ( ptype .eq. 2 ) type 90000,
0023 c      1 // kind // head // 'Triangular' // tail
0024 c      if ( ptype .eq. 3 ) type 90000,
0025 c      1 // kind // head // 'Increasing Ramp' // tail
0026 c      if ( ptype .eq. 4 ) type 90000,
0027 c      1 // kind // head // 'Decreasing Ramp' // tail
0028 c      if ( ptype .eq. 5 ) type 90000,
0029 c      1 // kind // head // 'Rectangular' // tail
0030 c
0031 c      return
0032 c
0033 c      90000 format ( $a )
0034 c
0035 c      end

```

18-Apr-1988 14:30:50

11-Mar-1988 08:31:27

```

0001      subroutine sigmat_estm( apars, num_ants, doa, num_doas, wavelen,
0002      1
0003      c
0004      c -----
0005      c
0006      c           This subroutine forms an estimate of the "signal matrix", whose
0007      c           columns are the "signal vectors" determined by subroutine SIGVAL, from
0008      c           estimated angles of arrival.
0009      c
0010      c           Parameters:
0011      c           -----
0012      c           apars = matrix whose first column is the vector of ranges (from
0013      c           the origin) of the antenna array elements, and
0014      c           whose second column is the vector of angles
0015      c           (from the "x" axis) of the antenna array
0016      c           elements (RADIANS). num_ants x 2. REAL*8.
0017      c           Note that the position of the ith antenna element is
0018      c           apars(i,1) * cos( apars(i,2) ) (x position),
0019      c           apars(i,1) * sin( apars(i,2) ) (y position).
0020      c           num_ants = number of antenna elements in the array. INTEGER*4
0021      c           doa = vector of estimated angles of arrival of the incoming
0022      c           waves. num_doas elements. REAL*8.
0023      c           num_doas = number of estimated angles of arrival. INTEGER*4.
0024      c           wavelen = wavelength of the incident waves. REAL*8. (Note
0025      c           that the units of wavelength are assumed to be
0026      c           those in which the array ranges are expressed)
0027      c           sigmat = computed signal matrix. num_ants rows, num_doas
0028      c           columns. COMPLEX*16.
0029      c
0030      c -----
0031      c
0032      c           implicit none
0033      c
0034      c           external sigval, dc_colfill
0035      c           external open_va, free_va, exit_va
0036      c
0037      c           integer*4 num_ants, num_doas
0038      c           real*8 apars( num_ants, 1 ), doa( num_doas ), wavelen
0039      c           complex*16 sigmat( num_ants, num_doas )
0040      c
0041      c           integer*4 dc_type/7/, linear(1), va_status
0042      c           character*80 va_msg
0043      c
0044      c           integer*4 sigvec_add, i
0045      c
0046      c           Make a place to receive computed signal vectors.
0047      c
0048      c           linear( 1 ) = num_ants
0049      c           call open_va( sigvec_add, linear, 1, dc_type, va_status, va_msg )
0050      c           call exit_va( va_status, va_msg )
0051      c
0052      c           Now compute signal vectors in turn, placing the results into
0053      c           the columns of sigmat.
0054      c
0055      c           do i = 1, num_doas
0056      c           call sigval( apars, num_ants, doa(i), wavelen, %val( sigvec_add ) )
0057      c           call dc_colfill( sigmat, num_ants, num_doas, i, %val( sigvec_add ) )

```

SIGMAT_ESTM

18-Apr-1988 14:30:50

11-Mar-1988 08:31:27

```

0058          end do
0059          c
0060          Discard work array and exit.
0061          c
0062          call free_va( sigvec_add, va_status, va_msg )
0063          call exit_va( va_status, va_msg )
0064          c
0065          return
0066          end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	225	PIC CON REL LCL SHR EXE RD NOW
1 \$PDATA	4	PIC CON REL LCL SHR NOEXE RD NOW
2 \$LOCAL	360	PIC CON REL LCL NOSHR NOEXE RD W
Total Space Allocated	589	

ENTRY POINTS

Address	Type	Name
0-00000000		SIGMAT_ESTM

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type	Name
2-00000054	1*4	DC_TYPE	2-00000060	1*4	I	AP-00000008	1*4	I
2-0000005C	1*4	SIGVEC_ADD	2-00000004		CHAR VA_MSG	2-00000058	1*4	I

ARRAYS

Address	Type	Name	Bytes	Dimensions
AP-00000004	R*8	APARS	**	(*, 1)
AP-0000000C	R*8	DDA	**	(*)
2-00000000	1*4	LINEAR	4	(1)
AP-00000018	C*16	SIGMAT	**	(*, *)

FUNCTIONS AND SUBROUTINES REFERENCED

Type	Name	Type	Name	Type	Name	Type	Name
	DC_COLFILL		EXIT_VA		FREE_VA		OPEN_VA

18-Apr-1988 14:31:
10-Sep-1987 08:58:

```

0001      subroutine sigmat_form( apars, num_ants, wpars, num_waves, sigmat
0002      c
0003      c-----
0004      c
0005      c          This subroutine forms the "signal matrix", whose columns are
0006      c          the "signal vectors" determined by subroutine SIGVAL.
0007      c
0008      c          apars = matrix whose first column is the vector of ranges (from the
0009      c          origin) of the antenna array elements, and whose second
0010      c          column is the vector of angles (from the "x" axis)
0011      c          of the antenna array elements (RADIANS). num_ants x
0012      c          REAL*8.
0013      c          Note that the position of the lth antenna element is
0014      c          apars(l,1) * cos( apars(l,2) ) (x position),
0015      c          apars(l,1) * sin( apars(l,2) ) (y position).
0016      c          num_ants = number of antenna elements in the array. INTEGER*4.
0017      c          wpars = matrix whose third column is the vector of angles (from the
0018      c          axis) of arrival of the incident waves, and whose
0019      c          fourth column is the vector of wavelengths of the incident
0020      c          waves. num_waves x 9. REAL*8. (Note that the units
0021      c          of wavelength are assumed to be those in which the
0022      c          array ranges are expressed.)
0023      c          num_waves = number of incident waves for which the signal matrix is
0024      c          to be formed. INTEGER*4.
0025      c          sigmat = computed signal matrix. num_ants rows, num_waves columns
0026      c          COMPLEX*16.
0027      c
0028      c-----
0029      c
0030      c          implicit none
0031      c
0032      c          external sigval, dc_colfill
0033      c          external open_va, free_va, exit_va
0034      c
0035      c          integer*4 num_ants, num_waves
0036      c          real*8 apars( num_ants, 1 ), wpars( num_waves, 1 )
0037      c          complex*16 sigmat( num_ants, num_waves )
0038      c
0039      c          integer*4 dc_type/7/, linear(1), va_status
0040      c          character*80 va_msg
0041      c
0042      c          integer*4 sigvec_add, i
0043      c
0044      c          Make a place to receive computed signal vectors.
0045      c
0046      c          linear( 1 ) = num_ants
0047      c          call open_va( sigvec_add, linear, 1, dc_type, va_status, va_msg )
0048      c          call exit_va( va_status, va_msg )
0049      c
0050      c          Now compute signal vectors in turn, placing the results in
0051      c          the columns of sigmat.
0052      c
0053      c          do i = 1, num_waves
0054      c             call sigval( apars, num_ants,
0055      c             1, wpars(i,3), wpars(i,9), %val( sigvec_add )
0056      c             call dc_colfill( sigmat, num_ants, num_waves, i, %val( sigvec_ad
0057      c          end do

```

SIGMAT_FORM

18-Apr-1988 14:31
10-Sep-1987 08:58

```

0058 c
0059 c          Discard work array and exit.
0060 c
0061 c          call free_va( sigvec_add, va_status, va_msg )
0062 c          call exit_va( va_status, va_msg )
0063 c
0064 c          return
0065 c          end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	248	PIC CON REL LCL SHR EXE RD
1 \$PDATA	4	PIC CON REL LCL SHR NOEXE RD
2 \$LOCAL	372	PIC CON REL LCL NOSHR NOEXE RD
Total Space Allocated		624

ENTRY POINTS

Address	Type	Name
0-00000000		SIGMAT_FORM

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type
2-00000054	1*4	DC_TYPE	2-00000060	1*4	I	AP-00000008@	I
2-0000005C	1*4	SIGVEC_ADD	2-00000004		CHAR VA_MSG	2-00000058	I

ARRAYS

Address	Type	Name	Bytes	Dimensions
AP-00000004@	R*8	APARS	**	(*, 1)
2-00000000	1*4	LINEAR	4	(1)
AP-00000014@	C*16	SIGMAT	**	(*, *)
AP-0000000C@	R*8	WPARS	**	(*, 1)

FUNCTIONS AND SUBROUTINES REFERENCED

Type	Name	Type	Name	Type	Name	Type	Name
	DC_COLFILL		EXIT_VA		FREE_VA		OPEN_VA

18-Apr-1988 14:31:25

9-Sep-1987 16:22:11

```

0001      subroutine sigval( apars, num_ants, arriv_angl, wavelen, sig_vec_val
0002      c
0003      c -----
0004      c
0005      c      This subroutine computes the value of the signal vector from
0006      c      array of receivers due to an incident wave, as a function of the angl
0007      c      of arrival of the incident wave.
0008      c
0009      c      Parameters:
0010      c      -----
0011      c      apars = matrix whose first column is the vector of ranges (fr
0012      c      the origin) of array antenna elements, and
0013      c      whose second column is the vector of angles
0014      c      (from the "x" axis) of array antenna elements
0015      c      (RADIANS). num_ants x 2. REAL*8.
0016      c      Note: the position of the lth antenna element is
0017      c      apars(i,1) * cos( apars(i,2) ) (x position),
0018      c      apars(i,1) * sin( apars(i,2) ) (y position).
0019      c      num_ants = number of elements in antenna array. INTEGER*4.
0020      c      arriv_angl = angle (from the "x" axis) of arrival for which t
0021      c      signal vector value is to be computed.
0022      c      Expressed in RADIANS. REAL*8.
0023      c      wavelen = wavelength of the incident wave. REAL*8. (Note th
0024      c      the units of wavelength must be the same as
0025      c      those in which the array_range(*) are
0026      c      expressed.)
0027      c      sig_vec_val = computed signal vector value. num_ants element
0028      c      COMPLEX*16. Each element is a complex value
0029      c      expressing the phase shift due to the positio
0030      c      of the corresponding array antenna element,
0031      c      based on the incident wavelength and the angl
0032      c      of arrival.
0033      c
0034      c -----
0035      c
0036      c      implicit none
0037      c
0038      c      include '(real_constants)'
0039      c
0040      c      intrinsic dcos, dsin, dcmplx
0041      c
0042      c      integer*4 num_ants
0043      c      real*8 apars( num_ants, 1 ), arriv_angl, wavelen
0044      c      complex*16 sig_vec_val( num_ants )
0045      c
0046      c      integer*4 cond_code, i
0047      c      real*8 phase, phase_rate
0048      c
0049      c      phase_rate = pi8t2 / wavelen
0050      c
0051      c      do i = 1, num_ants
0052      c          phase = phase_rate * apars(i,1) * dcos( apars(i,2) - arriv_angl )
0053      c          sig_vec_val(i) = dcmplx( dcos( phase ), -dsin( phase ) )
0054      c      end do
0055      c
0056      c      return
0057      c      end

```


18-Apr-1988 14:31:36
9-Sep-1987 16:53:14

```

0001      subroutine sigvec( num_waves, carr_camp, wpars, mtypes,
0002      1                      samp_num, samp_int, sig_vec
0003      c
0004      c-----
0005      c
0006      c          This subroutine computes the "signal-in-space" vector,
0007      c          consisting of the "complex amplitudes" (amplitudes and phases,
0008      c          expressed in "rectangular" coordinates) of the arriving waves for
0009      c          sample number samp_num.
0010      c
0011      c          Parameters:
0012      c          -----
0013      c          num_waves = INTEGER*4 number of waves impinging on the anter
0014      c                      array.
0015      c          carr_camp = num_waves-element COMPLEX*16 vector of carrier
0016      c                      complex amplitudes (from SNRs and "starting"
0017      c                      phases).
0018      c          wpars = num_waves x 9 REAL*8 matrix whose columns include:
0019      c                      4) vector of modulation cycle lengths, in
0020      c                          number of carrier cycles per modula-
0021      c                          cycle.
0022      c                      5) vector of modulation duty cycles (expres-
0023      c                          as fractions of modulation cycle
0024      c                          lengths, 0.0 < mod_duty(i) (<= 1.0)
0025      c                          (apply only to ramp and rectangular
0026      c                          modulation).
0027      c                      6) vector of "starting" modulation phases,
0028      c                          number of carrier cycles.
0029      c                      7) vector of amplitude modulation ratios
0030      c                          (maximum / minimum).
0031      c                      8) vector of phase modulation total phase
0032      c                          shifts (in RADIANS).
0033      c          mtypes = num_waves x 2 INTEGER*4 matrix whose first column
0034      c                      the vector of amplitude modulation types and
0035      c                      whose second column is the vector of phase
0036      c                      modulation types (see below).
0037      c          samp_num = INTEGER*4 number of the sample for which the comp
0038      c                      amplitudes are to be computed.
0039      c          samp_int = REAL*8 number of carrier cycles between samples.
0040      c          sig_vec = num_waves-element COMPLEX*16 output vector of comp
0041      c                      amplitudes.
0042      c
0043      c          Modulation types: 1 = sine, 2 = triangle, 3 = increasing r.
0044      c          4 = decreasing ramp, 5 = rectangular.
0045      c-----
0046      c
0047      c
0048      c          implicit none
0049      c          external modulated
0050      c
0051      c          complex*16 modulated
0052      c
0053      c          integer*4 num_waves, mtypes( num_waves, 1 )
0054      c          integer*4 samp_num
0055      c          real*8 wpars( num_waves, 1 ), samp_int
0056      c          complex*16 carr_camp( num_waves ), sig_vec( num_waves )
0057      c

```

SIGVEC

18-Apr-1988 14:31:

9-Sep-1987 16:53:

```

0058          integer*4 i
0059      c
0060          do i = 1, num_waves
0061              sig_vec(i) = modulated( i, num_waves, carr_camp, wpars, mtypes,
0062                  1
0063                  samp_num, samp_in'
0063          end do
0064      c
0065          return
0066      end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	184	PIC CON REL LCL SHR EXE RD
2 \$LOCAL	228	PIC CON REL LCL NOSHR NOEXE RD
Total Space Allocated	412	

ENTRY POINTS

Address	Type	Name
0-00000000		SIGVEC

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type
2-00000000	I*4	I	AP-00000004@	I*4	NUM_WAVES	AP-00000018@	R

ARRAYS

Address	Type	Name	Bytes	Dimensions
AP-00000008@	C*16	CARR_CAMP	**	(*)
AP-00000010@	I*4	MTYPES	**	(*, 1)
AP-0000001C@	C*16	SIG_VEC	**	(*)
AP-0000000C@	R*8	WPARS	**	(*, 1)

FUNCTIONS AND SUBROUTINES REFERENCED

Type	Name
C*16	MODULATED

18-Apr-1988 14:38:26

2-Apr-1988 17:00:57

```

0001      integer*4 function spawn_if( log_unit, prog_num )
0002      c
0003      c-----
0004      c
0005      c          This function reads (and locks) the record with primary key
0006      c          value prog_num in the indexed-organization file open on logical unit
0007      c          number log_unit.  If the specified record is locked when the read
0008      c          attempt is made, the function loops until the read (and lock) is
0009      c          successful.
0010      c          Once the desired record has been read and locked, the functi
0011      c          checks the "spawned" indication in the record and, if it is 0, spawn
0012      c          a subprocess which executes the command file named in the record, se
0013      c          the "spawned" indication to -1, and rewrites (and unlocks) the recor
0014      c          If the "spawned" indication is already -1, the function returns with
0015      c          taking any action.
0016      c          The value returned by the function is 0 if all operations
0017      c          succeeded (whether the subprocess was spawned or not), 1 if the read
0018      c          failed, 2 if the read succeeded but the spawn failed, and 3 if both
0019      c          read and spawn succeeded but the rewrite failed.
0020      c          The FORTRAN program MAKE_SPAWN_IDX can be used to create a f
0021      c          of the required format.  The text file
0022      c          6000$RT:[CHUCK.RESEARCH.FORTDIRJSPAWNED_DEF.TXT contains the
0023      c          definitions required to access such a file.  The parameters log_unit
0024      c          and prog_num are INTEGER*4.
0025      c
0026      c-----
0027      c
0028      c          implicit none
0029      c          external lib$spawn, lib$wait
0030      c          include '(system_symbols)'
0031      c          include 'SYS$LIBRARY:FORIOSDEF'
0032      c          include '6000$RT:[CHUCK.RESEARCH.FORTDIRJSPAWNED_DEF.TXT'
0033      c
0034      c          integer*4 lib$spawn
0035      c
0036      c          integer*4 log_unit, prog_num
0037      c
0038      c          integer*4 out_int
0039      c          character*1 uparrow( 3 ) / 27, 'E', 'A' /, spaces( 70 ) / 70* ' ' /
0040      c
0041      c          10000
0042      c          continue
0043      c          read (unit=log_unit,keyeq=prog_num,keyid=0,iostat=spawn_if)
0044      c          1          sp_num, sp_stat, sp_file, sp_proc
0045      c          if          ( spawn_if .eq. FOR$IOS_SPERECLC ) then          ! Record lo
0046      c          unlock (unit=log_unit)
0047      c          type *, 'Spawn: Record #', prog_num, ' locked.  Waiting 1 sec.',
0048      c          1          upar
0049      c          call lib$wait( 1.0 )
0050      c          type *, spaces, uparrow
0051      c          go to 10000
0052      c          else if ( spawn_if .ne. 0 ) then          ! Error rea
0053      c          unlock (unit=log_unit)
0054      c          spawn_if = 1
0055      c          return
0056      c          end if
0057      c
0058      c          if ( sp_stat .eq. 0 ) then          ! Batch program not started

```

```

SPAWN_IF                                     18-Apr-1988 14:38:2
                                              2-Apr-1988 17:00:5
0302      spawn_if = lib$spawn( , sp_file, '6000$RT:[CHUCK.PARAMS]SPAWN', ,
0303      1                                     out_int )
0304      if ( spawn_if .ne. SS$_NORMAL ) then
0305          unlock (unit=log_unit)
0306          spawn_if = 2
0307          return
0308      end if
0309      c
0310      sp_stat = -1
0311      rewrite (unit=log_unit,err=20000) sp_num, sp_stat, sp_file, sp_pr
0312      end if
0313      c
0314      unlock (unit=log_unit)
0315      spawn_if = 0
0316      return
0317      c
0318      20000      continue                                     ! Rewrite error.
0319      unlock (unit=log_unit)
0320      spawn_if = 3
0321      return
0322      c
0323      end

```

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	365	PIC CON REL LCL SHR EXE RD N
1 \$PDATA	72	PIC CON REL LCL SHR NOEXE RD N
2 \$LOCAL	300	PIC CON REL LCL NOSHR NOEXE RD
Total Space Allocated	737	

ENTRY POINTS

Address	Type	Name
0-00000000	1*4	SPAWN_IF

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type
AP-00000004@	1*4	LOG_UNIT	2-000000B4	1*4	OUT_INT	AP-00000008@	1*4
2-000000AC	1*4	SP_NUM	2-00000049	CHAR	SP_PROC	2-000000B0	1*4

18-Apr-1988 14:38:46
20-Feb-1988 18:22:34

```

0001      real*8 function spect_peak( iu, il, tol, p, apars, num_ants,
0002      1
0003      c
0004      c -----
0005      c
0006      c          This function returns the angle, in RADIANS, at which a peak
0007      c occurs in the MUSIC DOA spectrum, given an interval to search for th
0008      c peak. The method used is called the "golden section search" (see
0009      c Haykin, Simon, "Radar Array Processing for Angle of Arrival
0010      c Estimation", ARRAY SIGNAL PROCESSING, S. Haykin, Editor, Prentice-Ha
0011      c Inc., Englewood Cliffs, New, Jersey, 1985, pp. 213-215).
0012      c
0013      c Parameters:
0014      c -----
0015      c          il: REAL*8 lower limit of interval to search, in RADIANS.
0016      c          iu: REAL*8 upper limit of interval to search, in RADIANS.
0017      c          tol: REAL*8 value of desired resolution, in RADIANS. When
0018      c              interval being searched is reduced to <= tol
0019      c              radians, or when the search interval stops
0020      c              changing between iterations, the routine exit
0021      c              with a result of the midpoint of the current
0022      c              search interval.
0023      c          p: REAL*4 power to which to raise the terms used in
0024      c              accumulating a DOA spectrum sample (p = 2 fo
0025      c              "ordinary" MUSIC).
0026      c          apars: matrix whose first column is the vector of ranges (f
0027      c              the origin) of array antenna elements and wh
0028      c              second column is the vector of angles (from
0029      c              "x" axis) of array antenna elements (in
0030      c              RADIANS). num_ants x 2. REAL*8.
0031      c              Note: the position of the ith antenna element is
0032      c              apars(i,1) * cos( apars(i,2) ) (x position),
0033      c              apars(i,1) * sin( apars(i,2) ) (y position).
0034      c          num_ants: number of elements in antenna array. INTEGER*4.
0035      c          wavelen: wavelength of the incident wave. REAL*8. (Note t
0036      c              the units of wavelength must be the same as
0037      c              those in which the array_range(*) are
0038      c              expressed.)
0039      c          eigvec: num_ants x num_ants COMPLEX*16 matrix whose FIRST
0040      c              num_zeigs columns are the "noise" generalize
0041      c              eigenvectors defined in the MUSIC algorithm.
0042      c          num_zeigs: INTEGER*4 number of "noise" eigenvectors to be u
0043      c              in evaluating the MUSIC DOA measure.
0044      c
0045      c -----
0046      c
0047      c          implicit none
0048      c          external open_va, exit_va, free_va
0049      c          external music_power, sigval
0050      c
0051      c          real*8 music_power
0052      c
0053      c          integer*4 num_ants, num_zeigs
0054      c          real*4 p
0055      c          real*8 il, iu, tol, apars( num_ants, 1 ), wavelen
0056      c          complex*16 eigvec( num_ants, num_ants )
0057      c

```

SPECT_PEAK

18-Apr-1988 14:38:46

20-Feb-1988 18:22:34

```

0058      integer*4 steerv_add, linear(1), status, dc_type/7/
0059      real*8 u, l, a, b, fa, fb, lvl, oldlvl, tau/1.618033988749895/
0060      character*80 msg
0061      c
0062      linear(1) = num_ants
0063      call open_va( steerv_add, linear, 1, dc_type, status, msg )
0064      call exit_va( status, msg )
0065      c
0066      if ( iu .gt. il ) then
0067          u = lu
0068          l = ll
0069      else
0070          u = il
0071          l = lu
0072      end if
0073      lvl = u - l
0074      a = u - lvl / tau
0075      call sigval( apars, num_ants, a, wavelen, Xval( steerv_add ) )
0076      fa = music_power( eigvec, Xval( steerv_add ), num_ants, num_zelgs, p
0077      b = l + lvl / tau
0078      call sigval( apars, num_ants, b, wavelen, Xval( steerv_add ) )
0079      fb = music_power( eigvec, Xval( steerv_add ), num_ants, num_zelgs, p
0080      c
0081      oldlvl = tau * lvl
0082      do while ( ( lvl .gt. tol ) .and. ( lvl .lt. oldlvl ) )
0083          oldlvl = lvl
0084      c
0085          if ( fa .lt. fb ) then
0086              l = a
0087              lvl = u - l
0088              a = b
0089              fa = fb
0090              b = l + lvl / tau
0091              call sigval( apars, num_ants, b, wavelen, Xval( steerv_add ) )
0092              fb = music_power( eigvec, Xval( steerv_add ),
0093                  num_ants, num_zelgs, p )
0094          else
0095              u = b
0096              lvl = u - l
0097              b = a
0098              fb = fa
0099              a = u - lvl / tau
0100              call sigval( apars, num_ants, a, wavelen, Xval( steerv_add ) )
0101              fa = music_power( eigvec, Xval( steerv_add ),
0102                  num_ants, num_zelgs, p )
0103          end if
0104      end do
0105      c
0106      call free_va( steerv_add, status, msg )
0107      call exit_va( status, msg )
0108      c
0109      spect_peak = ( u + l ) / 2.0d0
0110      return
0111      end

```

18-Apr-1988 14:39:08 VA:
27-Mar-1988 15:10:03 SQ

```

0001      subroutine sq_vlgsvd( a, b, n, fuzz, eig, vec, condit, info )
0002      c
0003      c-----
0004      c
0005      c          This subroutine uses Van Loan's simplified algorithm for the
0006      c          generalized singular value decomposition (SVD) to compute the
0007      c          generalized eigenvalues and eigenvectors of the matrix pair (a, b).
0008      c
0009      c          Parameters
0010      c          -----
0011      c          a, b      n x n COMPLEX*16 UPPER TRIANGULAR, non-singular
0012      c                   matrices.
0013      c          n          INTEGER*4 number of rows / columns in a and b.
0014      c          fuzz      REAL*8 tolerance value ("closeness to zero").
0015      c          eig        n-element REAL*8 vector of generalized eigenvalues.
0016      c          vec        n x n COMPLEX*16 matrix of generalized eigenvectors
0017      c                   (columns of vec are eigenvectors).
0018      c          condit    REAL*8 estimate of the inverse of the condition number
0019      c                   of matrix a.
0020      c          info      INTEGER*4 success indicator. 0 if routine succeeded,
0021      c                   > 0 if a or b is singular, -1 if any
0022      c                   singular value has a significant
0023      c                   imaginary part, < -1 if any computed
0024      c                   singular values / vectors are in error.
0025      c
0026      c-----
0027      c
0028      c          implicit none
0029      c          external dc_matmpy, va_dc_trinv, va_dc_sqsvd, dc_matcopy, dc_vecmaxim
0030      c          external open_va, exit_va, eig_from_sv
0031      c
0032      c          real*8 dc_vecmaxim
0033      c
0034      c          integer*4 n, info
0035      c          real*8 fuzz, eig(n), condit
0036      c          complex*16 a(n,n), b(n,n), vec(n,n)
0037      c
0038      c          integer*4 status, linear(1), square(2), sv_err
0039      c          integer*4 bai_add, sv_add, lmt_add, ai_add, rmt_add
0040      c          integer*4 dr_type/5/, dc_type/7/
0041      c          character*80 msg
0042      c
0043      c          linear( 1 ) = n
0044      c          square( 1 ) = n
0045      c          square( 2 ) = n
0046      c
0047      c          Create working space.
0048      c
0049      c          call open_va( ai_add, square, 2, dc_type, status, msg )
0050      c          call exit_va( status, msg )
0051      c          call open_va( bai_add, square, 2, dc_type, status, msg )
0052      c          call exit_va( status, msg )
0053      c          call open_va( sv_add, linear, 1, dc_type, status, msg )
0054      c          call exit_va( status, msg )
0055      c          call open_va( lmt_add, square, 2, dc_type, status, msg )
0056      c          call exit_va( status, msg )
0057      c          call open_va( rmt_add, square, 2, dc_type, status, msg )

```

SQ_VLGSVD

18-Apr-1988 14:39:08 U.
27-Mar-1988 15:10:03 S

```

0058      call exit_va( status, msg )
0059      c
0060      c          Compute a inverse.
0061      c
0062      call dc_matcopy( a, n, n, %val( ai_add ) )
0063      call va_dc_trinv( %val( ai_add ), n, condit, info )
0064      if ( info .ne. 0 ) go to 10000
0065      c
0066      c          Do SVD of b * a-inverse.
0067      c
0068      call dc_matmpy( b, n, n, %val( ai_add ), n, %val( bai_add ) )
0069      call va_dc_sqsvd( %val( bai_add ), n, %val( sv_add ), %val( lmt_add ),
0070      1 %val( rmt_add ), sv_err )
0071      if ( dc_vecmaxim( %val( sv_add ), n ) .gt. fuzz ) info = -1
0072      if ( sv_err .gt. 0 ) info = -1 - sv_err
0073      c
0074      c          Compute eigenvalues and eigenvectors from a inverse and SVD of
0075      c          b * a-inverse.
0076      c
0077      call dc_matmpy( %val( ai_add ), n, n, %val( rmt_add ), n, vec )
0078      call eig_from_sv( %val( sv_add ), n, fuzz, eig )
0079      c
0080      c          Release working space and exit.
0081      c
0082      10000 continue
0083      call free_va( ai_add, status, msg )
0084      call exit_va( status, msg )
0085      call free_va( bai_add, status, msg )
0086      call exit_va( status, msg )
0087      call free_va( sv_add, status, msg )
0088      call exit_va( status, msg )
0089      call free_va( lmt_add, status, msg )
0090      call exit_va( status, msg )
0091      call free_va( rmt_add, status, msg )
0092      call exit_va( status, msg )
0093      c
0094      return
0095      end

```


SQ_VLGSVD

13-Apr-1988 14:39:08

27-Mar-1988 15:10:03

PROGRAM SECTIONS

Name	Bytes	Attributes
0 \$CODE	590	PIC CON REL LCL SHR EXE RD NOWRT
1 \$PDATA	8	PIC CON REL LCL SHR NOEXE RD NOWRT
2 \$LOCAL	648	PIC CON REL LCL NOSHR NOEXE RD WRT
Total Space Allocated		1246

ENTRY POINTS

Address	Type	Name
0-00000000		SQ_VLGSVD

VARIABLES

Address	Type	Name	Address	Type	Name	Address	Type	Name
2-00000070	I*4	AI_ADD	2-00000064	I*4	BAI_ADD	AP-0000001C@	R*8	CO
**	I*4	DR_TYPE	AP-00000010@	R*8	FUZZ	AP-00000020@	I*4	IN
2-0000000C	CHAR	MSG	AP-0000000C@	I*4	N	2-00000074	I*4	RM
2-00000068	I*4	SV_ADD	2-00000060	I*4	SV_ERR			

ARRAYS

Address	Type	Name	Bytes	Dimensions
AP-00000004@	C*16	A	**	(*, *)
AP-00000008@	C*16	B	**	(*, *)
AP-00000014@	R*8	EIG	**	(*)
2-00000000	I*4	LINEAR	4	(1)
2-00000004	I*4	SQUARE	8	(2)
AP-00000018@	C*16	VEC	**	(*, *)

LABELS

Address	Label
0-000001F3	10000

FUNCTIONS AND SUBROUTINES REFERENCED

Type	Name	Type	Name	Type	Name	Type	Name
	DC_MATCOPY		DC_MATMPY	R*8	DC_VECMAXIM		EIG_FROM_SV
	OPEN_VA		VA_DC_SQSV		VA_DC_TRINV		

18-Apr-1988 14:39:57
11-Nov-1987 09:13:34

```

0001      subroutine va_dc_qr( matrix, rows, cols, out_mat )
0002      c
0003      c-----
0004      c          This subroutine uses subroutines from LINPAK and the virtual
0005      c          array system (copyright 1987 by Dwight Day) to perform a QR
0006      c          decomposition of the rows x cols COMPLEX*16 matrix matrix. The indices
0007      c          rows and cols are INTEGER*4.
0008      c          On exit, the cols x cols COMPLEX*16 matrix out_mat contains
0009      c          the upper triangular result of the decomposition (the R part of the QR
0010      c          decomposition).
0011      c          Note that rows MUST BE GREATER THAN OR EQUAL TO cols. If this
0012      c          is not the case, the routine EXITS WITHOUT DOING ANY COMPUTATION!.
0013      c-----
0014      c
0015      c          implicit none
0016      c          external zqrdc, open_va, free_va, exit_va
0017      c
0018      c          integer*4 rows, cols
0019      c          complex*16 matrix(rows,cols), out_mat(cols,cols)
0020      c
0021      c          integer*4 i, j, status
0022      c          integer*4 dint_type/3/, dcmp_type/7/, linear(1)
0023      c          integer*4 work1, work2, work3
0024      c          character*80 msg
0025      c
0026      c          if ( rows .lt. cols ) return
0027      c
0028      c          linear(1) = cols
0029      c          call open_va( work1, linear, 1, dcmp_type, status, msg )
0030      c          call exit_va( status, msg )
0031      c          call open_va( work2, linear, 1, dint_type, status, msg )
0032      c          call exit_va( status, msg )
0033      c          call open_va( work3, linear, 1, dcmp_type, status, msg )
0034      c          call exit_va( status, msg )
0035      c
0036      c          call zqrdc( matrix, rows, rows, cols,
0037      c                   1, Xval( work1 ), Xval( work2 ), Xval( work3 ), 0 )
0038      c
0039      c          call free_va( work1, status, msg )
0040      c          call exit_va( status, msg )
0041      c          call free_va( work2, status, msg )
0042      c          call exit_va( status, msg )
0043      c          call free_va( work3, status, msg )
0044      c          call exit_va( status, msg )
0045      c
0046      c          do i = 1, cols
0047      c             do j = 1, cols
0048      c                if ( j .lt. i ) then
0049      c                   out_mat(i,j) = ( 0.0d0, 0.0d0 )
0050      c                else
0051      c                   out_mat(i,j) = matrix(i,j)
0052      c                end if
0053      c             end do
0054      c          end do
0055      c
0056      c          return
0057      c          end

```

18-Apr-1988 14:40:22 U
23-Jun-1987 08:56:18 U

```

0001      subroutine va_dc_sqsvd( mat, n, sv, left, right, info )
0002      c
0003      c-----
0004      c
0005      c          This subroutine uses subroutine ZSVDC from LINPAK and
0006      c          subroutines from the virtual array system (copyright 1987 by Dwight
0007      c          Day) to perform the singular value decomposition of the n x n
0008      c          COMPLEX*16 matrix mat. The index n is INTEGER*4.
0009      c          NOTE THAT THE INPUT MATRIX mat IS OVERWRITTEN.
0010      c          Results of the decomposition are returned in the n-element
0011      c          COMPLEX*16 vector sv (singular values) and the n x n COMPLEX*16
0012      c          matrices left (left-hand result matrix) and right (right-hand result
0013      c          matrix).
0014      c          The INTEGER*4 value info is 0 if the SVD succeeded, > 0 if some
0015      c          of the singular values and singular vectors were uncomputable.
0016      c-----
0017      c
0018      c
0019      c          implicit none
0020      c          external zsvdc, open_va, exit_va, free_va
0021      c
0022      c          integer*4 n, info
0023      c          complex*16 mat(n,n), sv(n), left(n,n), right(n,n)
0024      c
0025      c          integer*4 dcmp_type/7/, linear(1), status, errv_add, wk_add
0026      c          character*80 msg
0027      c
0028      c          linear(1) = n
0029      c          call open_va( errv_add, linear, 1, dcmp_type, status, msg )
0030      c          call exit_va( status, msg )
0031      c          call open_va( wk_add, linear, 1, dcmp_type, status, msg )
0032      c          call exit_va( status, msg )
0033      c
0034      c          call zsvdc( mat, n, n, n, sv, %val( errv_add),
0035      c                   1, left, n, right, n, %val( wk_add ), 11, info )
0036      c
0037      c          call free_va( errv_add, status, msg )
0038      c          call exit_va( status, msg )
0039      c          call free_va( wk_add, status, msg )
0040      c          call exit_va( status, msg )
0041      c
0042      c          return
0043      c          end

```

18-Apr-1988 14:39:31 VA
27-Mar-1988 14:52:52 VA

```
0001      subroutine va_dc_trinv( matrix, size, condit, panic )
0002      c
0003      c -----
0004      c
0005      c           This subroutine uses the routines ZTRCO and ZTRDI from LINPAK,
0006      c and subroutines from the Virtual Array system (copyright 1987 by Dwight
0007      c Day) to estimate the condition number and compute the inverse of the
0008      c size x size COMPLEX*16 UPPER TRIANGULAR matrix matrix. The index size
0009      c is INTEGER*4. The REAL*8 value condit is set to an estimate of the
0010      c inverse of the condition number of matrix. The INTEGER*4 value panic is
0011      c set to 0 if the inversion is successful, panic > 0 if matrix is
0012      c singular.
0013      c           NOTE that the inverse REPLACES the input matrix (except that
0014      c the strictly lower triangle of the matrix is never referenced).
0015      c
0016      c -----
0017      c
0018      c implicit none
0019      c external ztrco, ztrdi, open_va, exit_va, free_va
0020      c include '6000%RT:[CHUCK.RESEARCH.FORTDIR]VATYPES.TXT'
0021      c
0022      c integer*4 size, panic
0023      c real*8 condit
0024      c complex*16 matrix( size, size )
0025      c
0026      c integer*4 size_lin( 1 ), work_add, vstatus
0027      c character*80 msg
0028      c
0029      c size_lin(1) = size
0030      c call open_va( work_add, size_lin, 1, dc_type, vstatus, msg )
0031      c call exit_va( vstatus, msg )
0032      c
0033      c call ztrco( matrix, size, size, condit, %val( work_add ), 1 )
0034      c
0035      c call free_va( work_add, vstatus, msg )
0036      c call exit_va( vstatus, msg )
0037      c
0038      c call ztrdi( matrix, size, size, 0, 011, panic )
0039      c
0040      c return
0041      c
0042      c end
```

VITA

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