Discrete Convolution by Means of Forward and Backward Modeling

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Abstract-The standard methods of performing discrete convolution, that is, directly in the time domain or by means of the fast Fourier transform in the frequency domain, implicitly assume that the signals to be convolved are zero outside the observation intervals. Often this assumption produces undesirable end effects which are particularly severe when the signals are short in duration. This paper presents an approach to discrete convolution which obviates the zero assumption. The method is structurally similar to the Burg method [1], which estimates the autocorrelation coefficients of a series in a manner which does not require a predefinition of the behavior of the signal outside of the known interval. The basic principle of the present approach is that each term of the convolution is recursively determined from previous terms in a manner consistent with the optimal modeling of one signal into the other. The recursion uses forward and backward modeling together with the Morf et al. [2] algorithm for computation of the prediction error filter. The method is illustrated by application to the computation of the analytic signal and the derivative.

I. INTRODUCTION

THE mathematical definition of convolution requires L knowledge of the two functions to be convolved from $-\infty$ to $+\infty$. In the discrete case, the integral representation is approximated by an infinite summation and it is implicitly assumed that the signals are zero outside of the observation interval. Often this assumption is indeed valid, for example, in the case of transient signals, but frequently the zero extension leads to highly undesirable end effects, which are magnified in the case of short data sets. A particularly apt example of the deleterious effects of the zero extension assumption in the context of this paper is the computation of the power spectrum of harmonic processes. As is well known, the resolution of the periodogram estimate is highly dependent on the data interval, and for short intervals large frequency shifts occur. In order to obviate these effects, Burg [1] adapted the maximum entropy formalism [3] to spectral analysis with considerable success. Burg also contributed by developing an ingenious approach to the computation of the au-

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tocorrelation function, which is required in the maximum entropy formalism, which is not dependent on the usual assumption of a zero extension to the data.

This paper presents an approach to the computation of the discrete convolution of two functions which to a large extent minimizes the undesirable end effects caused by the constraint of a zero data extension. The convolution is formulated as a problem in the least squares modeling of one function into another. We show that each term of the discrete convolution may be obtained recursively using the Levinson [4] scheme by means of forward and backward modeling. These operators are determined efficiently using the Morf et al. algorithm [2] for computing the prediction error operator, PEO, from the covariance normal equations. The forward and backward covariance scheme for computing autoregressive parameters was introduced in [5] to obviate the need for a Toeplitz autocovariance matrix in the normal equations which implies a zero extension. A previous method which was developed to allow bandpass filtering of short data sets [6] used an autoregressive modeling of the signal and forward and backward prediction. This approach, particularly if the series is extended forwards and backwards using the Morf et al. [2] algorithm, has much to recommend it especially when the signal-to-noise ratio is high. The present approach is quite different in spirit in that it does not rely on the actual prediction of the time series.

The algorithm which is developed in this paper is applied to two problems which are often encountered in practice. These are the computation of the Hilbert transform, and consequently the analytic signal, and the computation of the derivative. The analytic signal in particular has recently found considerable application in both spectral analysis [7] and seismic data processing [8].

II. THEORY

A. Forward and Backward Modeling

Initially, we will formulate the problem of obtaining the forward and backward shaping operators of order j + 1, h_{j+1} , and f_{j+1} , respectively, for shaping a signal x_i of length *m* points into a desired signal d_i . For clarity, we first of all assume that x_i is zero outside of the observation interval. With this assumption, the associated autocovariance matrix is Toeplitz, and we may write the normal equations for forward and backward modeling in a compact form, which we call the expanded form, as follows.

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The normal equations for forward modeling of order j + 1

$$\boldsymbol{R}_{h,j+2} \begin{bmatrix} 1\\ \boldsymbol{h}_{j+1} \end{bmatrix} = \begin{bmatrix} \boldsymbol{r}_{h,0} & \boldsymbol{r}_{xd,j+1}^{T}\\ \boldsymbol{r}_{xdj+1} & \boldsymbol{R}_{xx,j+1} \end{bmatrix} \begin{bmatrix} 1\\ \boldsymbol{h}_{j+1} \end{bmatrix}$$
$$= \begin{bmatrix} \boldsymbol{E}_{h,j+1}\\ \boldsymbol{0}_{j+1} \end{bmatrix}, \qquad (1)$$

where

 $(1 h_{j+1}^T) = (1, h_{j+1,1}, \cdots, h_{j+1,j+1})$ is the forward modeling error operator of order j + 1,

$$0_{j+1} = (0_1, 0_2, \cdots, 0_{j+1})^T,$$

 $E_{h,j+1}$ = modeling energy associated with the

forward modeling error operator,

$$r_{h,0} = \sum_{t=0}^{m+j-1} d_t^2, \text{ energy of the desired signal,}$$

$$\boldsymbol{R}_{xx,j+1} = \sum_{t=0}^{T} \boldsymbol{x}_{j+1} \boldsymbol{x}_{j+1}^{T}$$
, Toeplitz autocovariance

matrix,

$$\mathbf{x}_{j+1} = (x_{i}, x_{i-1}, \cdots, x_{i-j})^{T},$$

$$\mathbf{r}_{xd,j+1} = \sum_{t=0}^{m+j-1} d_{t}\mathbf{x}_{j+1} = (r_{xd,0}, r_{xd,1}, \cdots, r_{xd,j})^{T}.$$

(2)

The normal equations for backward modeling of order j + 1

$$\boldsymbol{R}_{f,j+2} \begin{bmatrix} \boldsymbol{f}_{j+1} \\ 1 \end{bmatrix} = \begin{bmatrix} \boldsymbol{R}_{xx,j+1} & \boldsymbol{r}_{dx,j+1} \\ \boldsymbol{r}_{dx,j+1}^{T} & \boldsymbol{r}_{f,0} \end{bmatrix} \begin{bmatrix} \boldsymbol{f}_{j+1} \\ 1 \end{bmatrix} = \begin{bmatrix} \boldsymbol{0}_{j+1} \\ \boldsymbol{E}_{f,j+1} \end{bmatrix},$$
(3)

where

$$(f_{j+1}^T \quad 1) = (f_{j+1,j+1}, \cdots, f_{j+1,1}, 1)$$
 is the

backward modeling error operator

of order j + 1,

 $E_{f,j+1}$ = modeling energy associated with the

backward modeling error operator,

$$r_{f,0} = \sum_{t=0}^{m+j-1} d_{t-j}^{2}$$

$$r_{dx,j+1} = \sum_{t=0}^{m+j-1} d_{t-j} \mathbf{x}_{j+1}^{T}$$

$$= (r_{xd,-j}, \cdots, r_{xd,-1}, r_{xd,0})^{T}.$$
(4)

We note at this stage that the cross-correlation vectors given by (2) and (4) represent, respectively, a dislocation to the left and right of the signal d_t with respect to the signal x_t .

B. Least Squares Modeling and Discrete Convolution

In this section we show that each coefficient of the series which represents the discrete convolution of two signals is intimately related to the least squares shaping of one of the signals into the other. Let us allow the desired signal d_i to represent the time reverse of a particular filtering operator, which we designate as y_t , shifted to position k of x_t . In other words, $d_t = y_{-t+k}$ and clearly the vector $\mathbf{r}_{xd,j+1}$ now represents j + 1 terms of the convolution between y_t and x_t situated to the left of the initial term $r_{rd,0}$. On the other hand, in the case of the backward modeling operator, the shift between the two signals occurs in the opposite direction and this implies that the cross-correlation vector corresponds to convolution terms which are situated to the right of $r_{xd,0}$. Let us designate by \hat{c}_{k-j} and \hat{c}_{k+j} those terms of the convolution associated with the coefficients $r_{xd,j}$ and $r_{xd,-j}$, respectively. Substituting y_{-t+k} for d_t in (2) and y_{-t+k-j} for d_{t-j} in (4), we obtain for order j + 1

$$\hat{\boldsymbol{c}}_{\Theta,j+1} = \sum_{t=0}^{m+j-1} y_{k-t} \boldsymbol{x}_{j+1} = (\hat{c}_k, \, \hat{c}_{k-1}, \, \cdots, \, \hat{c}_{k-j})^T,$$
(5)

$$\hat{\boldsymbol{c}}_{\oplus,j+1} = \sum_{t=0}^{m+j-1} y_{k-j-t} \boldsymbol{x}_{j+1} = (\hat{c}_{k+j}, \, \hat{c}_{k+j-1}, \, \cdots, \, \hat{c}_k)^T.$$
(6)

The vectors expressed in (5) and (6) describe the 2j + 1 terms of the convolution between x_i and y_i which occur to the left and right of the central point \hat{c}_k , respectively, and which are expressed by

$$\hat{c}_{k-j}$$
 \hat{c}_{k-j+1} \cdots \hat{c}_k \cdots \hat{c}_{k+j-1} \hat{c}_{k+j} .

Thus far we have demonstrated the relationship which exists between the cross-correlation vector associated with the normal equations for the shaping or modeling filter and the terms of the discrete convolution of two signals. We now show, using the well-known expressions for the recursive solution of the normal equations (1) and (3), how the convolution products are related to the prediction error operators themselves.

The Levinson recursions for the forward and backward modeling operators of order j + 1 are

$$\begin{bmatrix} 1 \\ \boldsymbol{h}_{j+1} \end{bmatrix} = \begin{bmatrix} 1 \\ \boldsymbol{h}_{j} \\ 0 \end{bmatrix} + \boldsymbol{h}_{j+1,j+1} \begin{bmatrix} 0 \\ \boldsymbol{J}_{j}\boldsymbol{g}_{j} \\ 1 \end{bmatrix}, \quad (7)$$

and

$$\begin{bmatrix} \boldsymbol{f}_{j+1} \\ 1 \end{bmatrix} = \begin{bmatrix} \boldsymbol{0} \\ \boldsymbol{f}_{j} \\ 1 \end{bmatrix} + \boldsymbol{f}_{j+1,j+1} \begin{bmatrix} 1 \\ \boldsymbol{g}_{j} \\ 0 \end{bmatrix}, \quad (8)$$

where

$$\begin{pmatrix} 1 & g_j^T \end{pmatrix} = \begin{pmatrix} 1, g_{j,1}, \cdots, g_{j,j} \end{pmatrix}$$
, the prediction error

$$\boldsymbol{J}_{j} = \begin{bmatrix} \boldsymbol{0} & 1 \\ 1 & \\ 1 & \boldsymbol{0} \end{bmatrix}_{j \times j}.$$

C

The Toeplitz nature of the autocorrelation matrix implies that the forward and backward PEO's are equal. For the same reason, the error energies of forward and backward prediction are also equal and are represented by E_g . With this in mind, we can represent the normal equations in a compact form, as given in (9) and (10), which allow the recursive computation of the shaping operators. Derivation of (9) and (10) is presented in Appendix A.

$$\begin{bmatrix} E_{h,j} & \Delta_{h,j} \\ \Delta_{h,j} & E_{g,j} \end{bmatrix} \begin{bmatrix} 1 \\ h_{j+1,j+1} \end{bmatrix} = \begin{bmatrix} E_{h,j+1} \\ 0 \end{bmatrix}, \qquad (9)$$

$$\begin{bmatrix} E_{g,j} & \Delta_{f,j} \\ \Delta_{f,j} & E_{f,j} \end{bmatrix} \begin{bmatrix} f_{j+1,j+1} \\ 1 \end{bmatrix} = \begin{bmatrix} 0 \\ E_{f,j+1} \end{bmatrix}.$$
(10)

In the usual manner, we assume knowledge of the shaping operators and the respective errors at order j, and we compute the updated coefficients by means of (9), (10), (7), and (8). The terms $\Delta_{h,j}$ and $\Delta_{f,j}$ are obtained in terms of the cross-correlation vector and the PEO (Appendix), and may be written in terms of the vectors $\hat{c}_{\ominus,j+1}$ and $\hat{c}_{\oplus,j+1}$ as follows:

$$egin{aligned} &\Delta_{h,j} \,=\, \hat{c}_{k-j} \,+\, \hat{c}_{igodot,j}^T oldsymbol{J}_j oldsymbol{g}_j, \ &\Delta_{f,j} \,=\, \hat{c}_{k+j} \,+\, \hat{c}_{igodot,j}^T oldsymbol{g}_j. \end{aligned}$$

Rearranging the terms in these equations, we obtain

$$\hat{c}_{k-j} = -\hat{c}_{\ominus,j}^T \boldsymbol{J}_j \boldsymbol{g}_j + \Delta_{h,j}, \qquad (11)$$

$$\hat{c}_{k+j} = -\hat{c}_{\oplus,j}^T g_j + \Delta_{f,j}. \qquad (12)$$

The terms $\Delta_{h,j}$ and $\Delta_{f,j}$ may be written in terms of the updated coefficients and $E_{g,j}$ obtained from (9) and (10), and we may rewrite (11) and (12) as

$$\hat{c}_{k-j} = -\hat{c}_{\ominus,j}^T J_j g_j - h_{j+1,j+1} E_{g,j}, \qquad (13)$$

$$\hat{c}_{k+j} = -\hat{c}_{\oplus,j}^T g_j - f_{j+1,j+1} E_{g,j}.$$
 (14)

Equations (13) and (14) show that with knowledge of the PEO, the prediction error energy and the coefficients $h_{j+1,j+1}$ and $f_{j+1,j+1}$ for orders $j = 0, 1, \dots, p$, we can recursively determine the 2*p* convolution terms adjacent to \hat{c}_k . A schematic representation of (11) and (12) is pre-

sented below:

backward prediction

forward prediction

The above diagram clearly illustrates the principal which we use to determine the discrete convolution between signals x_t and y_t . The big dots • represent products. Each term is determined recursively from terms previously calculated or known in a manner which is consistent with the least error in shaping one signal into the other.

C. Discrete Convolution Shaping Algorithm

In the previous section, we derived expressions for the discrete convolution of two signals with the assumption that the data could be extended outside of the data interval with zeros. In this case, (13) and (14) give equivalent results to the conventional approach. In the general case, however, the problem is to avoid this restrictive extension. Using the above modeling approach, we now insist that the forward and backward modeling operators do not run off the data. The matrix which now takes part in the normal equations for order j + 1 take the form given in (15) and (16).

$$C_{h,j+2}\begin{bmatrix}1\\h_{j+1}\end{bmatrix} = \begin{bmatrix}c_{h,j+1} & c_{\ominus,j+1}^{T}\\c_{\ominus,j+1} & C_{xx,j+1}\end{bmatrix}\begin{bmatrix}1\\h_{j+1}\end{bmatrix} = \begin{bmatrix}E_{h,j+1}\\0_{j+1}\end{bmatrix}.$$
(15)

where

$$c_{h,j+1} = \sum_{t=j}^{m-1} d_t^2,$$

$$c_{\ominus,j+1} = \sum_{t=j}^{m-1} y_{k-1} x_{j+1} = (c_k, c_{k-1}, \cdots, c_{k-j})^T,$$

$$C_{xx,j+1} = \sum_{t=j}^{m-1} x_{j+1} x_{j+1}^T,$$

and

$$\boldsymbol{C}_{f,j+2}\begin{bmatrix}\boldsymbol{f}_{j+1}\\1\end{bmatrix} = \begin{bmatrix}\boldsymbol{C}_{xx,j+1} & \boldsymbol{c}_{\oplus,j+1}\\\boldsymbol{c}_{\oplus,j+1}^T & \boldsymbol{c}_{f,j+1}\end{bmatrix}\begin{bmatrix}\boldsymbol{f}_{j+1}\\1\end{bmatrix} = \begin{bmatrix}\boldsymbol{0}_{j+1}\\\boldsymbol{E}_{f,j+1}\end{bmatrix},$$
(16)

where

$$c_{f,j+1} = \sum_{t=j}^{m-1} y_{k-t+2}^{2},$$

$$c_{\oplus,j+1} = \sum_{t=j}^{m-1} y_{k-t+2} x_{j+1} = (c_{k+j}, c_{k+j-1}, \cdots, c_{k})^{T}.$$

We note that all the elements of the matrices in the above equations are functions of the model order. Further, $C_{xx,j+1}$ corresponds with the matrix in the normal equations associated with forward and backward prediction which, in expanded form, are given by

$$\boldsymbol{C}_{xx,j+1}\left[\begin{pmatrix}1\\\boldsymbol{a}_{j}\end{pmatrix}\begin{pmatrix}\boldsymbol{b}_{j}\\\boldsymbol{1}\end{pmatrix}\right] = \left[\begin{pmatrix}E_{a,j}\\\boldsymbol{0}_{j}\end{pmatrix}\begin{pmatrix}\boldsymbol{0}_{j}\\E_{b,j}\end{pmatrix}\right],\quad(17)$$

where

 $(1 \quad \boldsymbol{a}_{j}^{T}) = (1, a_{j,1}, \cdots, a_{j,j})$ the forward PEO, $(\boldsymbol{b}_{j}^{T} \quad 1) = (b_{j,j}, \cdots, b_{j,1}, 1)$ the backward PEO,

and where $E_{a,j}$ and $E_{b,j}$ correspond, respectively, to the forward and backward prediction error energies.

Once again, using the Levinson relation, we can express the shaping filters for order j + 1 assuming knowledge of the filters determined for order j in terms of the forward and backward PEO's,

$$\begin{bmatrix} 1\\ h_{j+1} \end{bmatrix} = \begin{bmatrix} 1\\ h_{j}\\ 0 \end{bmatrix} + h_{j+1,j+1} \begin{bmatrix} 0\\ b_{j}\\ 1 \end{bmatrix},$$
$$\begin{bmatrix} f_{j+1}\\ 1 \end{bmatrix} = \begin{bmatrix} 0\\ f_{j}\\ 1 \end{bmatrix} + f_{j+1,j+1} \begin{bmatrix} 1\\ a_{j}\\ 0 \end{bmatrix}.$$

In a manner analogous to the one in the previous section, we can write the normal equations in a compact form for forward and backward modeling as follows:

$$\begin{bmatrix} E_{h,j} & \Delta_{h,j} \\ \Delta_{h,j} & E_{b,j} \end{bmatrix} \begin{bmatrix} 1 \\ h_{j+1,j+1} \end{bmatrix} = \begin{bmatrix} E_{h,j+1} \\ 0 \end{bmatrix}, \quad (18)$$

$$\begin{bmatrix} E_{a,j} & \Delta_{f,j} \\ \Delta_{f,j} & E_{f,j} \end{bmatrix} \begin{bmatrix} f_{j+1,j+1} \\ 1 \end{bmatrix} = \begin{bmatrix} 0 \\ E_{f,j+1} \end{bmatrix}, \quad (19)$$

where

$$\Delta_{h,j} = \boldsymbol{c}_{\ominus,j}^T \boldsymbol{b}_j + c_{k-j} = -h_{j+1,j+1} E_{b,j}, \qquad (20)$$

$$\Delta_{f,j} = \boldsymbol{c}_{\oplus,j}^T \boldsymbol{a}_j + c_{k+j} = -f_{j+1,j+1} E_{a,j}.$$
(21)

Assuming that the first *j* terms of the convolution to the left and right of the initial coefficient \hat{c}_k are known, we can substitute (20) and (21) as well as a_j and b_j into (11) and (12) to obtain

$$\hat{c}_{k-j} = -\hat{\boldsymbol{c}}_{\Theta,j}^T \boldsymbol{b}_j - \boldsymbol{h}_{j+1,j+1} \boldsymbol{E}_{b,j}, \qquad (22)$$

$$\hat{c}_{k+j} = -\hat{\boldsymbol{c}}_{\oplus,j}^T \boldsymbol{a}_j - f_{j+1,j+1} E_{a,j}, \qquad (23)$$

or, equivalently, in terms of $\Delta_{h,j}$ and $\Delta_{f,j}$

$$\hat{c}_{k-j} = -\hat{c}_{\ominus,j}^{I} \boldsymbol{b}_{j} - \Delta_{h,j}, \qquad (24)$$

$$\hat{c}_{k+j} = -\hat{c}_{\oplus,j}^T a_j - \Delta_{f,j}. \qquad (25)$$

A diagrammatic representation of these relationships is shown below:

backward prediction

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forward prediction

Substituting the terms $\Delta_{h,j}$ and $\Delta_{f,j}$ from (20) and (21), we obtain the following equations for the desired convolutional algorithm:

$$\hat{c}_{k-j} = \left(\boldsymbol{c}_{\ominus,j} - \hat{\boldsymbol{c}}_{\ominus,j}\right)^T \boldsymbol{b}_j + c_{k-j}, \qquad (26)$$

$$\hat{\boldsymbol{c}}_{k+j} = \left(\boldsymbol{c}_{\oplus,j} - \hat{\boldsymbol{c}}_{\oplus,j}\right)^T \boldsymbol{a}_j + \boldsymbol{c}_{k+j}.$$
(27)

The first term on the right-hand side of (26) and (27) represents the inner product of the backward and forward PEO's, respectively, with the vectors which are the result of the difference between the *j* terms of the convolution previously calculated, and the *j* terms of the cross-correlation vector associated with the normal equations for forward and backward modeling of order j + 1. This first term represents the correction by which the coefficients c_{k-j} and c_{k+j} , which are computed as terms of the covariance matrix of the normal equations of order j + 1, must be adjusted in order to obtain the desired convolution coefficients. These correction terms are equal to zero when the signal x_t is zero extended outside of the data interval.

We note that the first j coefficients of $c_{\ominus,j+1}$ and the last j coefficients of $c_{\oplus,j+1}$ in (26) and (27), which are used at the j + 1 stage of the process, may be updated from the j coefficients determined at the previous stage using the expressions

$$\boldsymbol{c}_{\ominus,j+1} = \boldsymbol{c}_{\ominus,j} - y_{k-j+1}(x_{j-1}, x_{j-2}, \cdots, x_0)^T$$
, (28)

$$\boldsymbol{c}_{\oplus,j+1} = \boldsymbol{c}_{\oplus,j} - y_{-m+k+j} (x_{m-1}, x_{m-2}, \cdots, x_{m-j})^{\prime}.$$
(29)

We remark that the above expressions are structurally equivalent to the expressions in (11) and (12), or (13) and (14), and are also structurally equivalent to the form de-

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veloped by Burg [1] for the estimation of the autocorrelation function. Using (26) and (27) together with (28) and (29), we can recursively determine the 2 *j* convolution terms which occur to the left and right of the starting term \hat{c}_k and which are free of the usual assumptions of a zero extension to the data. The forward and backward PEO's which are required are efficiently determined using the algorithm developed by Morf *et al.* [2]. Our algorithm requires approximately $6m^2$ multiplications and additions.

III. APPLICATIONS

We consider the application of our convolution algorithm to two problems often encountered in practice. These are the computation of the Hilbert transform and the computation of the derivative. Both these functions are commonly determined either in the time domain by means of convolution with the appropriate operators or in the frequency domain. Results are of course identical if care is taken that the frequency domain computation represents discrete rather than circular convolution.

Example 1—The Discrete Hilbert Transform: In general, the discrete Hilbert transform may be computed either by convolution with the Hilbert operator or by the 90° phase splitter method [9]. For comparison purposes, we have adopted the convolutional approach. Fig. 1(a) shows the input signal which consists of two decaying sinusoids each represented as $x(t) = \exp(-\alpha_j t) \sin(2\pi t/T_j + \theta_j)$ where $\alpha_1 = 0.004$, $T_1 = 40 \text{ ms}$, $\theta_1 = 40^\circ$, and $\alpha_2 = 0.003$, $T_2 = 75 \text{ ms}$, and $\theta_2 = 20^\circ$. Fig. 1(b) and (c) shows, respectively, the envelopes of the input signal determined by means of the normal approach and by means of the algorithm developed in this paper. Fig. 1(d) and (e) shows the errors between the theoretical envelope and the envelopes shown in (b) and (c).

Example 2--The Derivative: The second example which we consider is that of the computation of the derivative of a signal. The input signal which we consider is the same as shown in Fig. 1(a). Fig. 2(a) and (b) illustrates the derivative computed using the standard convolutional approach and the recursive least squares approach, respectively. Fig. 2(c) and (d) shows the respective errors.

IV. DISCUSSION

The problem of convolutional end effects is a ubiquitous one. We have developed an algorithm which, unlike the approach presented in [6], does not require the actual prediction of the signal. The present technique models the input signal into the time reverse of a particular filter operator. The modeling is performed recursively using forward and backward modeling operators which are determined from the covariance normal equations using the efficient Morf *et al.* [2] algorithm for computing the PEO's. The algorithm is structurally similar to the Burg method [1] of determining the autocovariance without requiring the assumption of a zero extension. The initial



Fig. 1. (a) Input signal consists of two decaying sinusoids each represented as $x(t) = \exp(-\alpha_i t) \sin(2\pi t/T_i + \theta_i)$ where $\alpha_1 = 0.004$, $T_1 = 40$ ms, $\theta_1 = 40^\circ$, $\alpha_2 = 0.003$, $T_2 = 75$ ms, and $\theta_2 = 20^\circ$. (b) Envelope of the signal in (a) obtained using the normal frequency domain approach. (c) Envelope of the signal in (b) obtained using the algorithm developed in this paper. (d) Error between the true envelope and that shown in (b). (e) Error between the true envelope and that shown in (c).



Fig. 2. (a) Derivative of the trace shown in Fig. 1(a) computed using the standard method in the frequency domain. (b) Derivative of the trace shown in (a) computed using the algorithm developed in this paper. (c) Error between the true derivative and that shown in (a). (d) Error between the true derivative and that shown in (b).

convolution term, \hat{c}_k , plays a fundamental role in the process, and the fidelity of the final result depends on \hat{c}_k in the same manner as Burg's autocovariance estimate depends on the initial zero lag value. We illustrate various examples of the use of our algorithm and, in particular, we show the marked improvement in the obtained convolution for the case of various filtering operations involving decaying harmonics. An interesting aspect of the method which we have presented relates to the convolution of signals which may be modeled as AR processes with zero innovation. In this case, for a particular order

j, the forward and backward prediction errors, $E_{a,j}$ and $E_{b,j}$, respectively, will be zero and the 2j + 1 terms previously calculated may be extended indefinitely to the right and to the left of the computed series. The expressions for this particular case are immediately obtained from (22) and (23). These equations show that the forward and backward prediction operators, $-a_j$ and $-b_j$, respectively, which allow the exact prediction of the signal for positive and negative times, also allow the exact prediction of the convolution to the left and to the right of the previous 2j + 1 convolution terms. This result demonstrates that the convolution of AR signals depends only on the forward and backward prediction operators and those convolution terms which are used to initialize the process.

Another interesting aspect to the approach presented here occurs when we let the desired signal equal the input time series. In this case, the output of the algorithm is the autocorrelation of the input and, as before, the filters $-a_j$ and $-b_j$ may be used to extend the autocorrelation function into positive and negative times to obtain the estimates $r_{xx,j}$ and $r_{xx,-j}$ for order j. In order to ensure the nonnegative definite character of the autocorrelation function, these estimates may be averaged in a suitable manner as suggested in [10].

APPENDIX

We will assume knowledge of the modeling error operator of order j, $(1 \quad \boldsymbol{h}_{j}^{T})$, and the modeling error energy, $E_{h,j}$, which are related in the usual manner shown below:

$$\begin{bmatrix} \mathbf{r}_{h,0} & \mathbf{r}_{xd,j}^T \\ \mathbf{r}_{xd,j} & \mathbf{R}_{xx,j} \end{bmatrix} \begin{bmatrix} 1 \\ \mathbf{h}_j \end{bmatrix} = \begin{bmatrix} \mathbf{E}_{h,j} \\ \mathbf{0}_j \end{bmatrix}, \quad (A.1)$$

where the terms in the equation have been defined previously.

We also assume knowledge of the PEO and of the prediction error energy of order *j* which are related according to

$$\boldsymbol{R}_{xx,j} \begin{bmatrix} \boldsymbol{J}_{j} \boldsymbol{g}_{j} \\ 1 \end{bmatrix} = \begin{bmatrix} \boldsymbol{0}_{j} \\ \boldsymbol{E}_{g,j} \end{bmatrix}.$$
(A.2)

Let $Q_{h,j+1}$ represent the quadratic form associated with the normal equations of order j + 1 for forward modeling

$$\boldsymbol{Q}_{h,j+1} = \begin{bmatrix} 1 & \boldsymbol{h}_{j+1}^T \end{bmatrix} \boldsymbol{R}_{h,j+2} \begin{bmatrix} 1 \\ \boldsymbol{h}_{j+1} \end{bmatrix}. \quad (A.3)$$

Substituting $(1 \quad h_{j+1}^T)$ given by the Levinson recursion, we obtain for (A.3) [11]

$$Q_{h,j+1} = \begin{bmatrix} 1 & h_{j+1,j+1} \end{bmatrix} \begin{bmatrix} 1 & \boldsymbol{h}_j^T & 0 \\ 0 & \boldsymbol{g}_j^T \boldsymbol{J}_j & 1 \end{bmatrix}$$
$$\cdot \boldsymbol{R}_{h,j+2} \begin{bmatrix} 1 & 0 \\ \boldsymbol{h}_j & \boldsymbol{J}_j \boldsymbol{g}_j \\ 0 & 1 \end{bmatrix} \begin{bmatrix} 1 \\ h_{j+1,j+1} \end{bmatrix}.$$

$$Q_{h,j+1} = \begin{bmatrix} 1 & h_{j+1,j+1} \end{bmatrix} \begin{bmatrix} E_{h,j} & \Delta_{h,j} \\ \Delta_{h,j} & E_{g,j} \end{bmatrix} \begin{bmatrix} 1 \\ h_{j+1,j+1} \end{bmatrix},$$
(A.4)

where $\Delta_{h,j}$ is determined from either of the expressions in (A.5)

$$\Delta_{h,j} = \boldsymbol{r}_{xd,j} + \boldsymbol{r}_{xx,j}^T \boldsymbol{J}_j \boldsymbol{h}_j, \qquad (A.5a)$$

$$\Delta_{h,j} = \mathbf{r}_{xd,j} + \mathbf{r}_{xd,j}^T \mathbf{J}_j \mathbf{g}_j, \qquad (A.5b)$$

with

$$\mathbf{r}_{xx,j} = (r_{xx,1}, r_{xx,2}, \cdots, r_{xx,j})^{T}.$$

Minimizing $Q_{h,i+1}$ with respect to $h_{i+1,i+1}$ results in

$$\begin{bmatrix} \Delta_{h,j} & E_{g,j} \end{bmatrix} \begin{bmatrix} 1 \\ h_{j+1,j+1} \end{bmatrix} = 0.$$
 (A.6)

Substituting (A.6) into (A.4) results in the minimum error which is given by

$$\begin{bmatrix} E_{g,j} & \Delta_{h,j} \end{bmatrix} \begin{bmatrix} 1 \\ h_{j+1,j+1} \end{bmatrix} = E_{h,j+1}. \quad (A.7)$$

The union of (A.6) and (A.7) allows the compact representation which we have used in the text

$$\begin{bmatrix} E_{h,j} & \Delta_{h,j} \\ \Delta_{h,j} & E_{g,j} \end{bmatrix} \begin{bmatrix} 1 \\ h_{j+1,j+1} \end{bmatrix} = \begin{bmatrix} E_{h,j+1} \\ 0 \end{bmatrix}.$$
 (A.8)

A derivation of (10) in the text for backward modeling proceeds in an analogous manner.

We emphasize that the procedure for obtaining the compact form (A.8) is not dependent on the Toeplitz character of the autocovariance matrix. Thus, the solution of the normal equations of order j + 1 which contain a symmetric matrix may be obtained by means of the linear combination of solutions which satisfy forward and backward systems of order j. We explore this fact in a forth-coming paper [12].

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