

SIO 209: Signal Processing for Ocean Sciences Class 15

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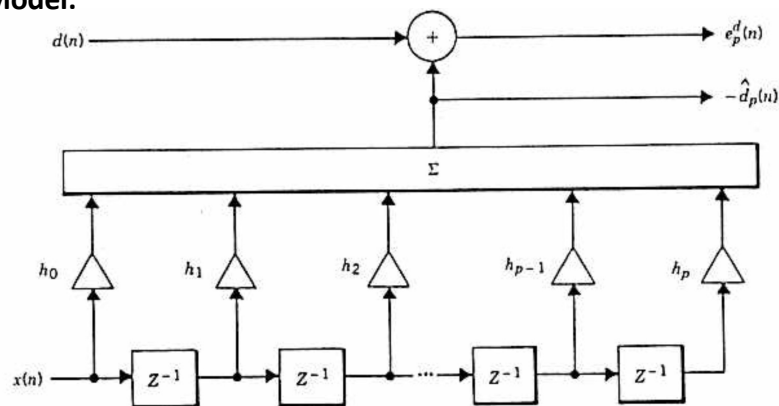


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Wiener Filtering and Least Squares Problem

- **General Filter Model:**



- **Goal:** Find FIR filter coefficients $h_0 \dots h_p$ such that the output of the FIR filter, $-\hat{d}_p[n]$, is as similar as possible to $d[n]$

D. DeFatta, J. Lucas, and W. Hodgkiss, "Digital Signal Processing." Wiley, 1988

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Wiener Filtering and Least Squares Problem

- We want to filter the time series $x[n]$ such that it yields an estimate of the time series $d[n]$
- The estimation error is denoted by

$$e^d[n] = d[n] + \sum_{k=0}^p h_k x[n-k]$$

- Vector notation of estimation error

$$e^d[n] = d[n] + \mathbf{h}^T \mathbf{x} \quad \mathbf{h} = [h_0 \ h_1 \ \cdots \ h_p]^T \quad \mathbf{x} = [x[n] \ x[n-1] \ \cdots \ x[n-p]]^T$$

- The optimality criteria for estimation are given by

$$\min_{\mathbf{h}} \mathbb{E} \left\{ |e^d[n]|^2 \right\} \quad \text{Wiener Filtering Problem} \qquad \min_{\mathbf{h}} \sum_n |e^d[n]|^2 \quad \text{Least Squares Problem}$$

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Wiener Filtering Problem

- For the Wiener filtering problem, we introduce the scalar ϕ_{00}^d , the vector \mathbf{g} , and the matrix as Φ

$$\begin{aligned} \phi_{00}^d &= \mathbb{E} \{ d[n] d^*[n] \} \\ \mathbf{g} &= [g_0 \ g_1 \ \cdots \ g_p]^T \end{aligned} \qquad \Phi = \begin{bmatrix} \phi_{00} & \cdots & \phi_{0p} \\ \vdots & & \vdots \\ \phi_{p0} & \cdots & \phi_{pp} \end{bmatrix} = \begin{bmatrix} \phi_{00} & \cdots & \phi_{0p} \\ \vdots & & \vdots \\ \phi_{0p}^* & \cdots & \phi_{pp} \end{bmatrix}$$

where we have used $g_j = \mathbb{E} \{ d[n] x^*[n-j] \}$ and $\phi_{kj} = \mathbb{E} \{ x[n-k] x^*[n-j] \}$

- The Wiener filtering problem follows a stochastic formulation and optimizes the filter in an expected value sense

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Least Squares Problem

- Alternatively, the least-square problem follows a deterministic formulation and optimizes for specific signal sequences
- Here, the statistics of the signals are computed from the data, i.e.,

$$\phi_{00}^d = \sum_n d[n] d^*[n] \quad g_j = \sum_n d[n] x^*[n-j] \quad \phi_{kj} = \sum_n x[n-k] x^*[n-j]$$

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Optimal Solution and Orthogonality Principle

- To solve for the unknown vector \mathbf{h} , we write the squared error as follows

$$|e^d[n]|^2 = d^*[n]d[n] + d^*[n]\mathbf{h}^T \mathbf{x} + \mathbf{h}^H \mathbf{x}^* d[n] + \mathbf{h}^H \mathbf{x}^* \mathbf{h}^T \mathbf{x}$$

- By making use of ϕ_{00}^d , \mathbf{g} , and Φ , the mean squared error can be obtained as

$$E_p^d(\mathbf{h}) = E\{|e^d[n]|^2\} = \phi_{00}^d + \mathbf{g}^H \mathbf{h} + \mathbf{h}^H \mathbf{g} + \mathbf{h}^H \Phi^T \mathbf{h}$$

- Minimizing $E_p^d(\mathbf{h})$ with respect to \mathbf{h} yields

$$0 = \mathbf{g} + \Phi^T \mathbf{h} \quad \text{or} \quad \Phi^T \mathbf{h} = -\mathbf{g}$$

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Optimal Solution and Orthogonality Principle

- The optimum coefficient vector of the Wiener filter is thus obtained as $\hat{\mathbf{h}} = -(\Phi^T)^{-1} \mathbf{g}$
- By using this solution in the expression of the mean-squared error, we obtain

$$E_p^d(\hat{\mathbf{h}}) = \phi_{00}^d - \mathbf{g}^H (\Phi^T)^{-1} \mathbf{g}$$

- Finally, let's compute the expectation of the product of the error and the complex conjugate of the data, i.e.,

$$\begin{aligned} E\{e^d[n] \mathbf{x}^*\} &= E\{d[n] \mathbf{x}^* + \mathbf{h}^T \mathbf{x} \mathbf{x}^*\} \\ &= \mathbf{g} + \Phi^T \mathbf{h} \end{aligned}$$

- It is straightforward to see that for $\mathbf{h} = \hat{\mathbf{h}}$, we have $E\{e^d[n] \mathbf{x}^*\} = 0$

→ **Orthogonality Principle of Linear Estimation:** The error of an optimum linear estimator is always orthogonal to the data

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Recursive Implementation of the Wiener Filter

- When the time series $x[n]$ is stationary, the matrix Φ has a Toeplitz structure, i.e., the elements on each diagonal are identical
- This property can be expressed mathematically as $\phi_{k,j} = \phi_{k-j}$
- The **Levinson–Durbin algorithm** exploits the structure to avoid direct computation of Φ^{-1}
- In particular, $\hat{\mathbf{h}}_p$ is recursively calculated from $\hat{\mathbf{h}}_{p-1}$ (see details in Chapter 11 of [DeFatta et al., 1988] and Chapter 11 of [Oppenheim & Schaffer, 2009])

D. DeFatta, J. Lucas, and W. Hodgkiss, "Digital Signal Processing." Wiley, 1988

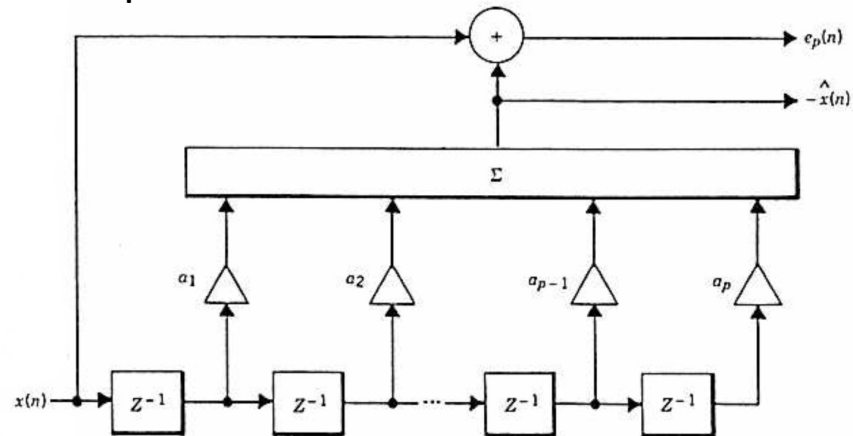
A. Oppenheim and R. Schaffer, "Discrete Time Signal Processing." Prentice Hall, 2009

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One Step Prediction Filter

- Special Case: One Step Forward Prediction Filter



It is straightforward to see that the one step forward prediction filter used for high resolution spectral analysis is a special case of a Wiener filter