EE3731C: Signal Processing Methods

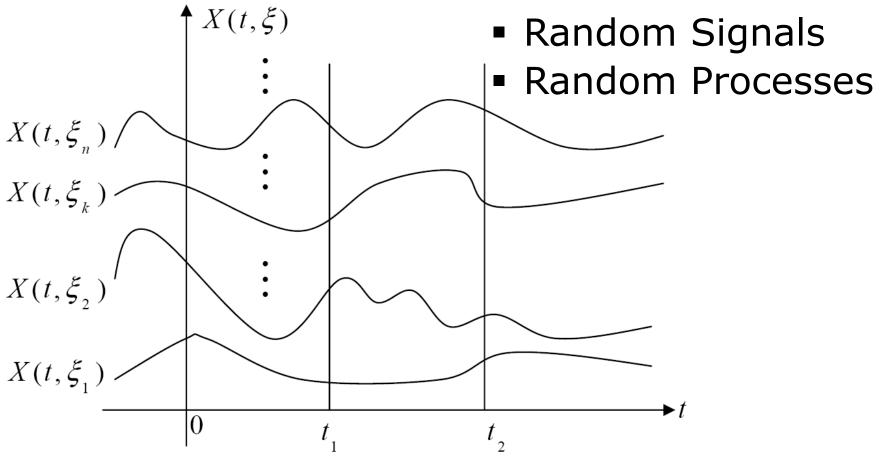
Lecture II-4: Random Signals II



Outline

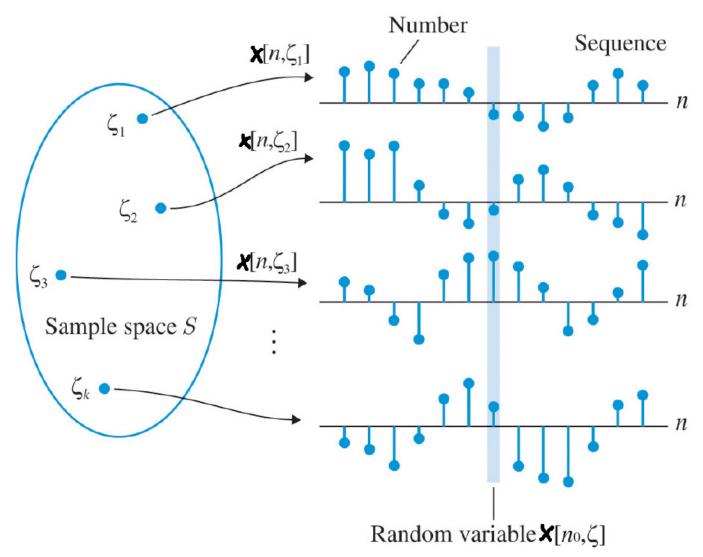
- Stochastic Processes
- Statistical Independence
- Linear Time-Invariant System
- Random Signals and Linear Systems
- Power Spectrum
- Cross Spectrum
- Wiener Filtering

Stochastic Processes (Revisit)



Can be completely specified by the collection of joint CDF/PDF among the random variables $\{X(t_1), X(t_2), \dots, X(t_n)\}$ for any set of $\{t_1, t_2, \dots, t_n\}$ and any order n.

Stochastic Processes



Statistical Independence

Two processes X(t) and Y(t) are independent if any two vectors of time samples, one from each process, are independent. That is,

 $[X(t_1), X(t_2), \cdots X(t_n)]$ and $[Y(t_1), Y(t_2), \cdots Y(t_n)]$ are independent random vectors.

 If X(t) and Y(t) are independent then they are uncorrelated (the reverse is not always true).

Linear Time-Invariant System

- Let x[n], x₁[n], and x₂[n] be inputs to a linear system and let y[n], y₁[n], and y₂[n] be their corresponding outputs.
- A linear system satisfies
 - -Additivity: $x_1[n] + x_2[n] \Rightarrow y_1[n] + y_2[n]$
 - -Homogeneity: $\alpha x[n] \Rightarrow \alpha y[n]$ for any constant α
- Let x[n] be the input to time-invariant system and y[n] be its corresponding output. Then, $x[n-m] \Rightarrow y[n-m]$, for any integer m
- Examples: $y_1[n] = (x[n])^2$ $y_2[n] = x[Mn]$

Linear Time-Invariant System

- Any linear time-invariant system (LTI)
 system can be uniquely characterized by its
 - -Impulse response: response of system to an impulse
 - -Frequency response: response of system to a complex exponential $e^{j2\pi f}$ for all possible frequencies f
 - -Transfer function: Laplace transform of impulse response
- Given one of the three, we can find other two provided that they exist.

Discrete-time Convolution

- Output y[n] for input x[n]
- Any signal can be decomposed into sum of discrete impulses
- Apply linear properties
- Apply shift-invariance
- Apply change of variables

$$y[n] = T\{x[n]\}$$

$$y[n] = T \left\{ \sum_{m=-\infty}^{\infty} x[m] \delta[n-m] \right\}$$

$$y[n] = \sum_{m=-\infty}^{\infty} x[m] T\{\delta[n-m]\}$$

$$y[n] = \sum_{m=-\infty}^{\infty} x[m]h[n-m]$$

$$y[n] = \sum_{m=-\infty}^{\infty} h[m]x[n-m]$$

$$h[n]$$
 Averaging filter impulse response $0 \quad 1 \quad 2 \quad 3$

$$\Rightarrow y[n]$$

$$y[n] = h[0] x[n] + h[1] x[n-1]$$

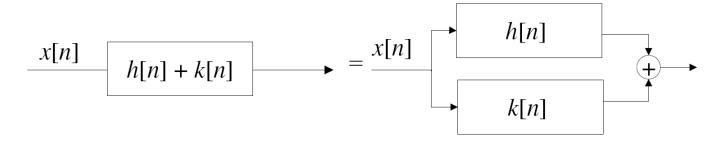
= $(x[n] + x[n-1]) / 2$

Properties

■ Commutative: x[n]*h[n] = h[n]*x[n]

$$x[n] \qquad \qquad \bullet = \frac{h[n]}{x[n]} \qquad \bullet$$

■ Distributive: x[n]*(h[n]+k[n])=x[n]*h[n]+x[n]*k[n]



■ Associative: x[n]*(h[n]*k[n])=(x[n]*h[n])*k[n]

$$x[n] \qquad h[n] * k[n] \qquad \longrightarrow = \frac{x[n]}{h[n]} \qquad h[n] \qquad \longrightarrow$$

Linear Difference Equations

- Discrete-time LTI systems can be characterized by difference equations.
- For example,

$$y[n] = (1/2) y[n-1] + (1/4) y[n-2] + x[n]$$

 Taking z-transform of difference equation gives description of system in z-domain.

Z-transform

 A polynomial representation of a sequence {x(n)} in terms of a complex-valued variable z⁻¹:

$$X(z) = \sum_{n=-\infty}^{\infty} x(n)z^{-n}$$

If X(z) = B(z)/A(z)
B(z), A(z): polynomials in z, then z_i such that B(z_i) = 0 are zeros of X(z), and
p_j such that A(p_j) = 0 are poles of X(z).

 Frequency domain representation of {x(n)} (discrete time Fourier transform, DTFT):

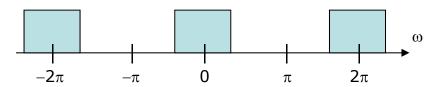
$$X(e^{j\omega}) = X(z)|_{z=e^{j\omega}} = \sum_{n=-\infty}^{\infty} x(n)e^{-j\omega n}$$

Convolution formula:

$$y(n) = x(n) * h(n)$$

 $\Leftrightarrow Y(z) = X(z)H(z)$

- Digital Frequency Axis:
 - Periodic, period = 2π



Transfer Function

A general Nth order difference equation

$$y[n] + a_1y[n-1] + \dots + a_Ny[n-N] = b_0x[n] + b_1x[n-1] + \dots + b_Nx[n-N]$$

Take Z-transform on both sizes

$$(1 + a_1 z^{-1} + \dots + a_{N-1} z^{-(N-1)} + a_N z^{-N}) Y[z] = (b_0 + b_1 z^{-1} + \dots + b_{N-1} z^{-(N-1)} + b_N z^{-N}) X[z]$$

Transfer function

$$H[z] = \frac{Y[z]}{X[z]} = \frac{b_0 + b_1 z^{-1} + \dots + b_{N-1} z^{-(N-1)} + b_N z^{-N}}{1 + a_1 z^{-1} + \dots + a_{N-1} z^{-(N-1)} + a_N z^{-N}}$$

Relation between h[n] and H[z]

- Either one can be used to describe an LTI system
 - Having one is equivalent to having the other since they are a z-transform pair
 - -By definition, impulse response h[n] is

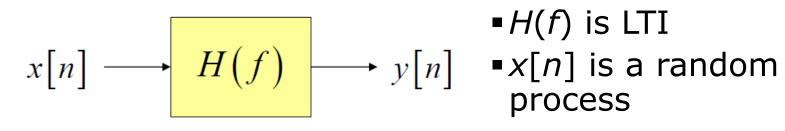
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y[n] = h[n] when x[n] = \delta[n]

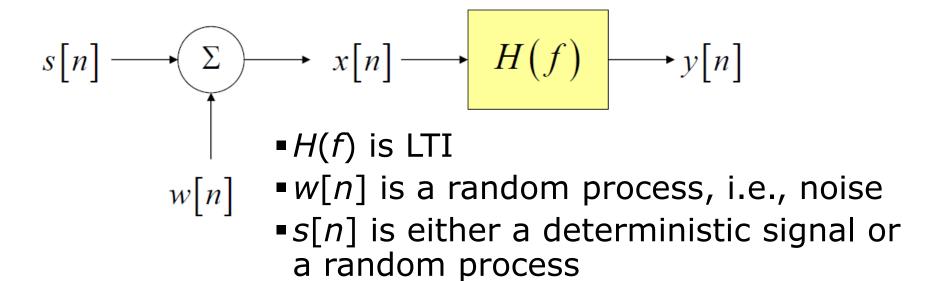
Z\{h[n]\} = H[z]Z\{\delta[n]\} \Rightarrow H[z] = H[z] \cdot 1

h[n] \Leftrightarrow H[z]
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 Since discrete-time signals can be built up from unit impulses, knowing the impulse response or the transfer function completely characterizes the LTI system.

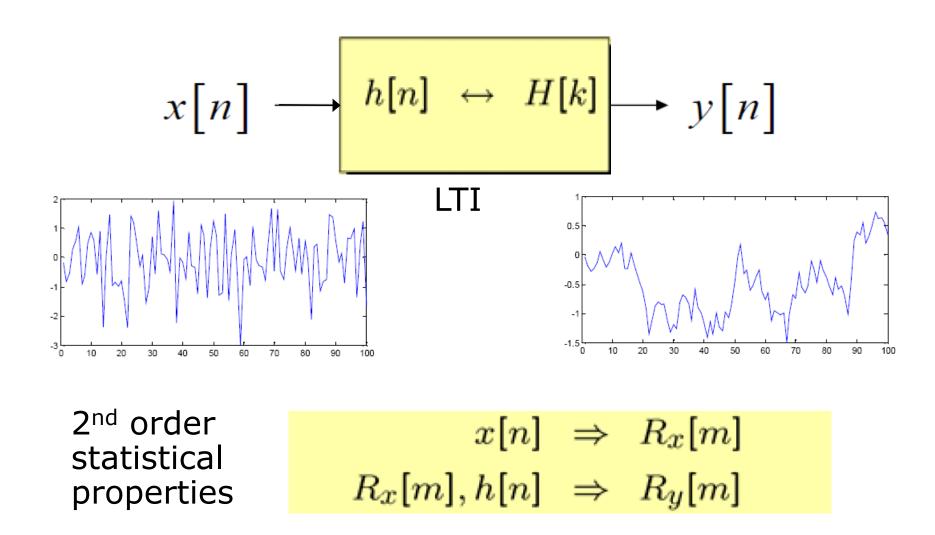
Random Processes & LTI Systems





What can we say about y[n]?

Random Sequences & LTI Systems



Response of an LTI System

Given a WSS ergodic random process, x[n], and LTI system with impulse response h[n] we can compute the time-average of the output y[n].

$$x[n] \longrightarrow h[n] \leftrightarrow H[k] \longrightarrow y[n]$$
LTI

$$\begin{split} m_y &= E\{y[n]\} \\ &= E\Big\{\sum_{k=-\infty}^{\infty} h[k]x[n-k]\Big\} = \sum_{k=-\infty}^{\infty} h[k]E\{x[n-k]\} \\ &= \sum_{k=-\infty}^{\infty} h[k]m_x = m_x \sum_{k=-\infty}^{\infty} h[k] = m_x H[0] \end{split}$$

Response of an LTI System

Given a WSS ergodic random process, x[n], and LTI system with impulse response h[n] we can compute the autocorrelation of the output y[n].

$$R_{y}[m] = E\{y[n]y[n+m]\}$$

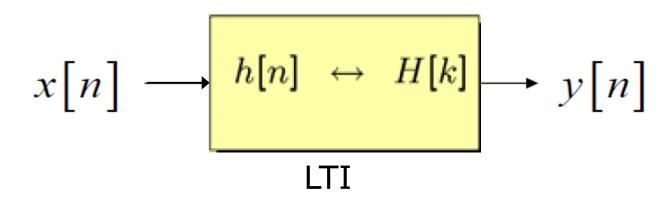
$$= E\{\sum_{k=-\infty}^{\infty} h[k]x[n-k] \sum_{l=-\infty}^{\infty} h[l]x[n+m-l]\}$$

$$= \sum_{k=-\infty}^{\infty} \sum_{l=-\infty}^{\infty} h[l]h[k]E\{x[n-k]x[n+m-l]\}$$

$$= \sum_{k=-\infty}^{\infty} \sum_{l=-\infty}^{\infty} h[l]h[k]R_{x}[m-l+k]$$

$$= R_{x}[m] * h[m] * h[-m]$$

Stochastic Process - Filtering



Filtering corresponds to convolution of the autocorrelation

$$R_{y}[m] = R_{x}[m] * h[m] * h[-m]$$



Power Spectral Density

The autocorrelation of the system response function is computed as the convolution of the h[m] with a time-reversed version h[-m].

$$R_h[m] = h[-m] * h[m]$$

It can be computed in the Fourier domain:

$$h[-m] * h[m] = \mathcal{F}^{-1} \{H^*(f)H(f)\}$$
$$= \mathcal{F}^{-1} \{|H(f)|^2\}$$
$$R_h[m] \leftrightarrow |H(f)|^2$$

Power Spectrum

The power spectrum of a random process is defined as the Fourier transform of the autocorrelation function:

$$R_x[m] \leftrightarrow S_x(f)$$

$$S_{x}(f) = \sum_{k=-\infty}^{\infty} R_{x}[k]e^{-j2\pi fk}$$

$$S_x(e^{j\omega}) = \sum_{k=-\infty}^{\infty} R_x[k]e^{-j\omega k}$$

The output power spectrum is given by:

$$S_y(f) = |H(f)|^2 S_x(f)$$
 $S_y(e^{j\omega}) = |H(e^{j\omega})|^2 S_x(e^{j\omega})$

$$S_y(e^{j\omega}) = |H(e^{j\omega})|^2 S_x(e^{j\omega})$$

$$R_{y}[m] = R_{x}[m] * h[m] * h[-m]$$

White Noise Sequences

White noise sequences: WSS processes whose autocorrelation function is:

$$R_w[m] = \langle w[n]w[n+m] \rangle$$
$$= \sigma_w^2 \delta[m]$$

where

$$\langle w[n] \rangle = 0$$

 $\langle w[n]^2 \rangle = \sigma_w^2$

Why is this called "white" noise?

- The white noise sequence is uncorrelated from sample to sample.
- > All i.i.d. sequences satisfy this constraint.
- ➤ It only requires the samples to be uncorrelated, but not independent.

Power Spectrum: White Noise

Autocorrelation function

$$R_w[m] = \langle w[n]w[n+m] \rangle$$
$$= \sigma_w^2 \delta[m]$$

Its power spectrum is a constant equal to the variance:

$$S_w(f) = \sigma_w^2$$

- White noise has equal power at all frequencies.
- ➤ By analogy with white light which is a uniform mixture of all visible frequencies.

Power Spectrum Estimation

Direct estimate of the power spectrum from the first N observations of a realization x[n]:

$$\widehat{S}_{x}(f) = \frac{1}{N} |X_{N}(f)|^{2}$$

$$= \frac{1}{N} \left| \sum_{n=0}^{N-1} x[n] e^{-j2\pi f n} \right|^{2}$$

- It does not converge to $S_x(f)$ as N grows large. \odot
- The variance of the estimate does not converge to 0 for increasing N.

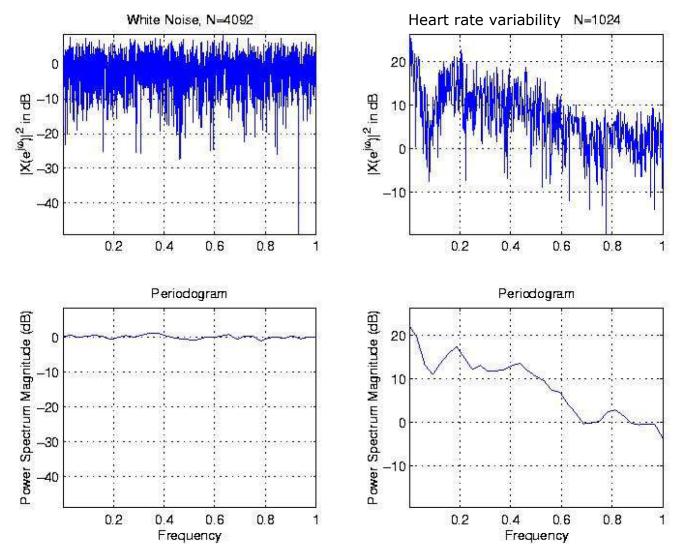
Power Spectrum Estimation

Averaging Periodogram can be used to smooth neighboring frequencies: Compute Fourier transform for window of size *N* and average over *M* windows.

$$\widehat{S}_x(f) = \frac{1}{NM} \sum_{m=0}^{M-1} |X_{N,M}(f)|^2$$

The estimate converges to the true Power Spectrum as *M* and *N* grow large.

Power Spectrum: Examples



Compute FT for window of size *N/K* and average over *K* windows.

Cross-correlation of Input and Output

$$x[n] \longrightarrow h[n] \leftrightarrow H[k] \longrightarrow y[n]$$
LTI

$$y[n] = \sum_{l=-\infty}^{\infty} h[l]x[n-l]$$

$$R_{xy}[m] = E\{x[n]y[n+m]\}$$

$$= E\{x[n]\sum_{l=-\infty}^{\infty} h[l]x[n+m-l]\}$$

$$= \sum_{l=-\infty}^{\infty} h[l]E\{x[n]x[n+m-l]\}$$

$$= \sum_{l=-\infty}^{\infty} h[l]R_x[m-l] = h[m]*R_x[m]$$

$$S_{xy}(z) = H(z)S_x(z)$$
$$S_{xy}(e^{j\omega}) = H(e^{j\omega})S_x(e^{j\omega})$$

Estimating System Response

If the input to the system is a white-noise sequence:

$$R_{wy}[m] = h[m] * R_w[m]$$

$$= h[m] * \sigma_w^2 \delta[m]$$

$$= \sigma_w^2 h[m]$$

then the 2nd order statistics of the output process can be used to estimate the system response function (i.e., system identification).

Cross-correlation of Input and Output

$$x[n] \longrightarrow h[n] \leftrightarrow H[k] \longrightarrow y[n] \quad R_{yx}[m] = E\{y[n]x[n+m]\}$$

$$= E\left\{\sum_{l=-\infty}^{\infty} h[l]x[n-l]x[n+m]\right\}$$

$$= \sum_{l=-\infty}^{\infty} h[l]x[n-l]x[n+m]$$

$$= \sum_{l=-\infty}^{\infty} h[l]E\{x[n-l]x[n+m]\}$$

$$= \sum_{l=-\infty}^{\infty} h[l]R_x[m+l]$$

$$= \sum_{l=-\infty}^{\infty} h[l]R_x[m+l]$$

$$= \sum_{l=-\infty}^{\infty} h[-(-l)]R_x[m-(-l)]$$

$$= \sum_{l=-\infty}^{\infty} h[-k]R_x[m-k] = h[-m] * R_x[m]$$

Cross Spectrum

The cross spectrum of the signals x[n] and y[n] is the Fourier transform of the cross-correlation function:

$$S_{xy}(f) \stackrel{\triangle}{=} \sum_{k=-\infty}^{\infty} R_{xy}[k] e^{-j2\pi fk}$$

Relationships:

$$R_{xy}[m] = h[m] * R_x[m] \to S_{xy}(f) = H(f) S_x(f)$$

 $R_{yx}[m] = R_{xy}[-m] \to S_{yx}(f) = S_{xy}(-f) = S_{xy}^*(f)$

When the input is zeromean, white noise: $S_{wy}(f) = \sigma_w^2 H(f)$

Power Spectrum: Summary

$$S_{x}(f) = \sum_{k=-\infty}^{\infty} R_{x}[k]e^{-j2\pi fk}$$

$$R_{xy}[n] \leftrightarrow S_{xy}(f)$$

$$h[n] \leftrightarrow H(f)$$

$$R_{x}[n] \leftrightarrow S_{x}(f)$$

$$S_y(f) = |H(f)|^2 S_x(f)$$

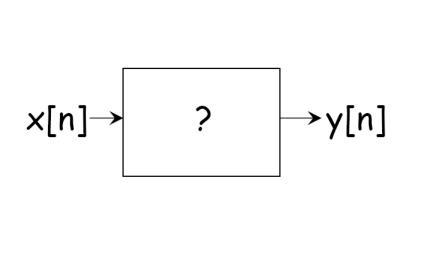
$$R_{xy}[n] = \sum_{k=-\infty}^{\infty} h[k] R_x[n-k]$$
$$= h[n] * R_x[n]$$

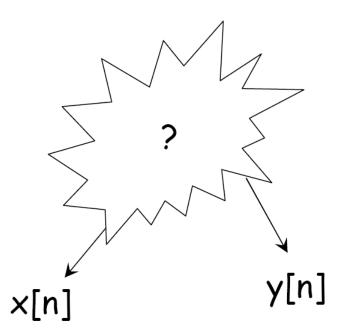
$$H(f) = \frac{S_{xy}(f)}{S_x(f)}$$

Each frequency is independent of all other frequencies.

Wiener Filtering

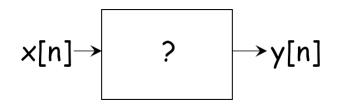
Suppose we have two jointly WSS random processes whose joint second-order statistics are known, and we want to estimate one from an observation of the other.





Wiener Filter: Example

Given two realizations of random processes, y[n] and x[n], how to construct a filter which predicts y[n] from observations of x[n]?



Minimize the mean square $\times[n]$? $\rightarrow y[n]$ error (MSE) between y[n] and the prediction $\hat{y}|n|$

$$\langle e[n]^2 \rangle = \langle (y[n] - \hat{y}[n])^2 \rangle$$

$$= \langle (y[n] - h[n] * x[n])^2 \rangle$$

$$= \langle \left(y[n] - \sum_k h[k] x[n-k] \right)^2 \rangle$$

$$= \sigma_y^2 - 2 \sum_k h[k] R_{xy}[k] + \sum_k \sum_l h[k] h[l] R_x[k-l]$$

Wiener Filter: Example

The error is a quadratic function of the filter coefficients h[k]. Taking the first order derivative of the MSE with respect to h[k] and setting it to zero, we obtain a system of linear equations.

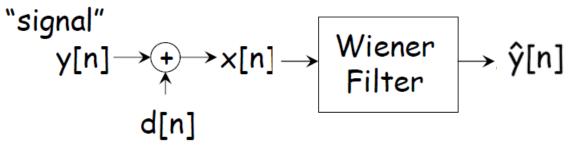
$$R_{xy}[k] = \sum_{l} h[l] R_x[k-l]$$

Known as Wiener-Hopf or Yule-Walker equations.

If we restrict to Nth-order FIR filters, then it becomes

$$\begin{bmatrix} R_{x}[0] & R_{x}[1] & \cdots & R_{x}[N-1] \\ R_{x}[1] & R_{x}[0] & & R_{x}[N-2] \\ \vdots & \vdots & \ddots & \vdots \\ R_{x}[N-1] & R_{x}[N-2] & \cdots & R_{x}[0] \end{bmatrix} \begin{bmatrix} h[0] \\ \vdots \\ h[N-1] \end{bmatrix} = \begin{bmatrix} R_{xy}[0] \\ \vdots \\ R_{xy}[N-1] \end{bmatrix}$$

Wiener Filter for Noise Removal



Uncorrelated noise:

"noise"

$$R_{yd}[n] = 0 \leftrightarrow S_{yd}(f) = 0$$

$$H(f) = \frac{S_{xy}(f)}{S_x(f)}$$
$$= \frac{S_y(f)}{S_y(f) + S_d(f)}$$

$$SNR(f) \triangleq \frac{S_{y}(f)}{S_{d}(f)}$$

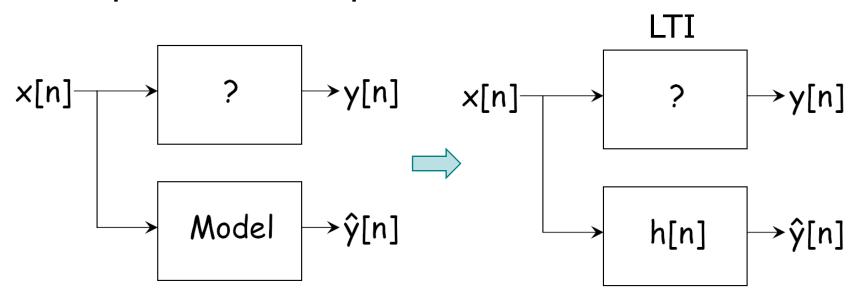
$$H(f) = \frac{S_{y}(f)}{S_{y}(f) + S_{d}(f)}$$

$$= \frac{SNR(f)}{SNR(f) + 1}$$

 $SNR(f) \gg 1$: $H(f) \approx 1$ $SNR(f) \ll 1$: $H(f) \approx 0$

Wiener Filter for System Identification

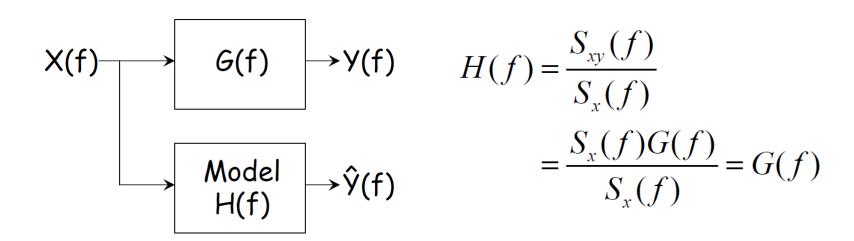
System Identification: Estimate a model of an unknown system based on observations of inputs and outputs.



Find the LTI system that best predicts the output from the input.

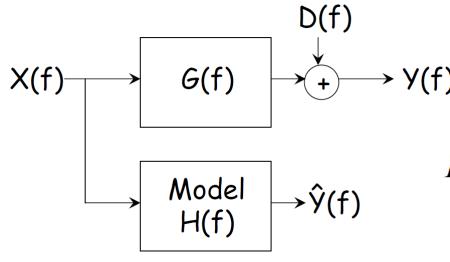
Example: LTI System

If the unknown system is LTI, we hope that Wiener filtering recovers that LTI system exactly.



Example: LTI System + Noise

We can only observe y[n] which is the output of the LTI system + noise



We can still recover the LTI system exactly!

d[n] is uncorrelated with x[n].

$$H(f) = \frac{S_{xy}(f)}{S_x(f)}$$

$$= \frac{S_x(f)G(f) + S_{xd}(f)}{S_x(f)}$$

$$= \frac{S_x(f)G(f)}{S_x(f)} = G(f)$$

Example: Spectral Factorization

Suppose that we know the autocorrelation function $R_x[m]$ of the input x[n] but do not have access to x[n] and therefore cannot determine the cross-correlation $R_{xy}[m]$ with the output y[n], but can determine the output autocorrelation $R_y[m]$.

If
$$R_x[m] = \delta[m]$$
, then $R_y[m] = h[m] * h[-m]$

$$S_{y}[z] = H[z]H[z^{-1}]$$

Additional assumptions or constraints, for instance on the stability and causality of the system and its inverse, may allow one to recover H(z) from knowledge of $H(z)H(z^{-1})$.

Example: Causal Wiener Filter

$$\begin{bmatrix} R_{x}[0] & R_{x}[1] & \cdots & R_{x}[N-1] \end{bmatrix} h[0] \\ R_{x}[1] & R_{x}[0] & R_{x}[N-2] & \vdots \\ \vdots & \vdots & \ddots & \vdots \\ R_{x}[N-1] & R_{x}[N-2] & \cdots & R_{x}[0] \end{bmatrix} h[0] \\ h[N-1] \end{bmatrix} = \begin{bmatrix} R_{xy}[0] \\ \vdots \\ R_{xy}[N-1] \end{bmatrix}$$

If $R_{\downarrow}[m] = \delta[m]$, then the system is simplified to

$$\begin{bmatrix} 1 & 0 & \cdots & 0 \\ 0 & 1 & & 0 \\ \vdots & \vdots & \ddots & \vdots \\ 0 & 0 & \cdots & 1 \end{bmatrix} h \begin{bmatrix} 0 \\ \vdots \\ h[N-1] \end{bmatrix} = \begin{bmatrix} R_{xy}[0] \\ \vdots \\ R_{xy}[N-1] \end{bmatrix}$$

Causal Wiener filter:
$$h[n] = \begin{cases} R_{xy}[n] & n \ge 0 \\ 0 & n < 0 \end{cases}$$

Wiener Filter: Summary

- A Wiener filter finds the MMSE (minimum mean squared error) estimate of one random process as a linear function of another random process.
- Applications include noise removal and system identification.
- How do we get R_x and R_{xy} ?
 - Sample statistics for auto- and crosscorrelations
 - Periodogram estimates of power spectral densities

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- Bertrand Delgutte. Course materials for HST.582J / 6.555J / 16.456J, Biomedical Signal and Image Processing, Spring2007. MIT OpenCourseWare (http://ocw.mit.edu), Massachusetts Institute of Technology.