

Lesson 1

Optimal Signal Processing

Optimal signalbehandling

LTH

September 2013

**Statistical Digital Signal Processing
and Modeling,
Hayes, M:**

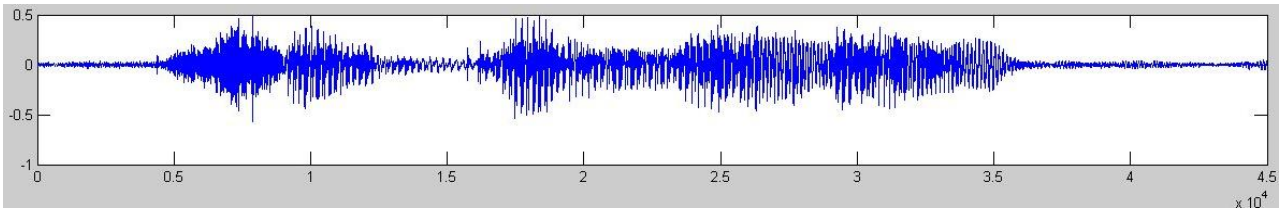
John Wiley & Sons, 1996. ISBN 0471594318

Nedelko Grbic

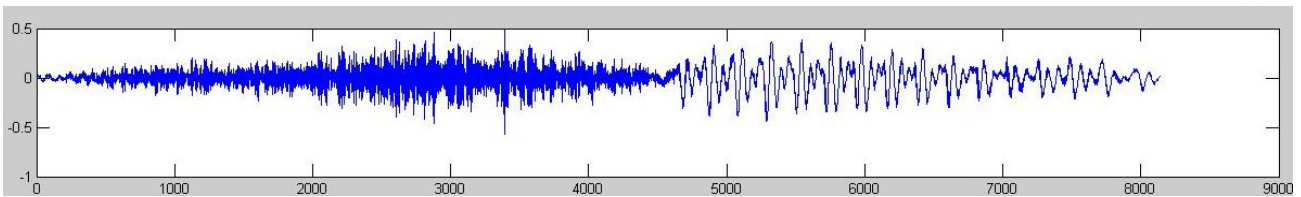
(Mtrl from Bengt Mandersson)

Department of
Electrical and Information Technology, Lund University
Lund University

The sound of 'Signalbehandling'



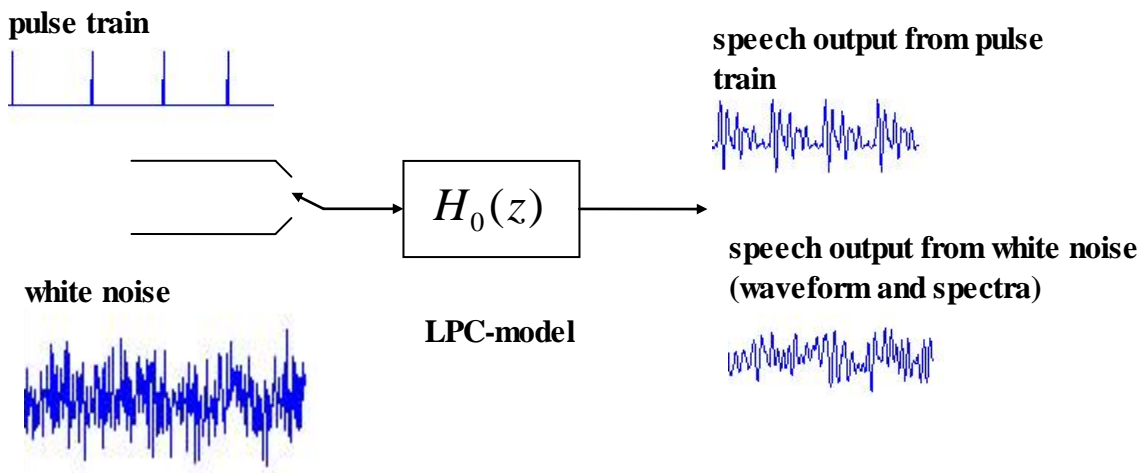
's' 'i' 'g' 'n' '.....'



's' 'i'
noise harmonic signal

How can this be generated as output from a linear filter?
Determine the filter and the input signal.

LPC model of syntetic sound production

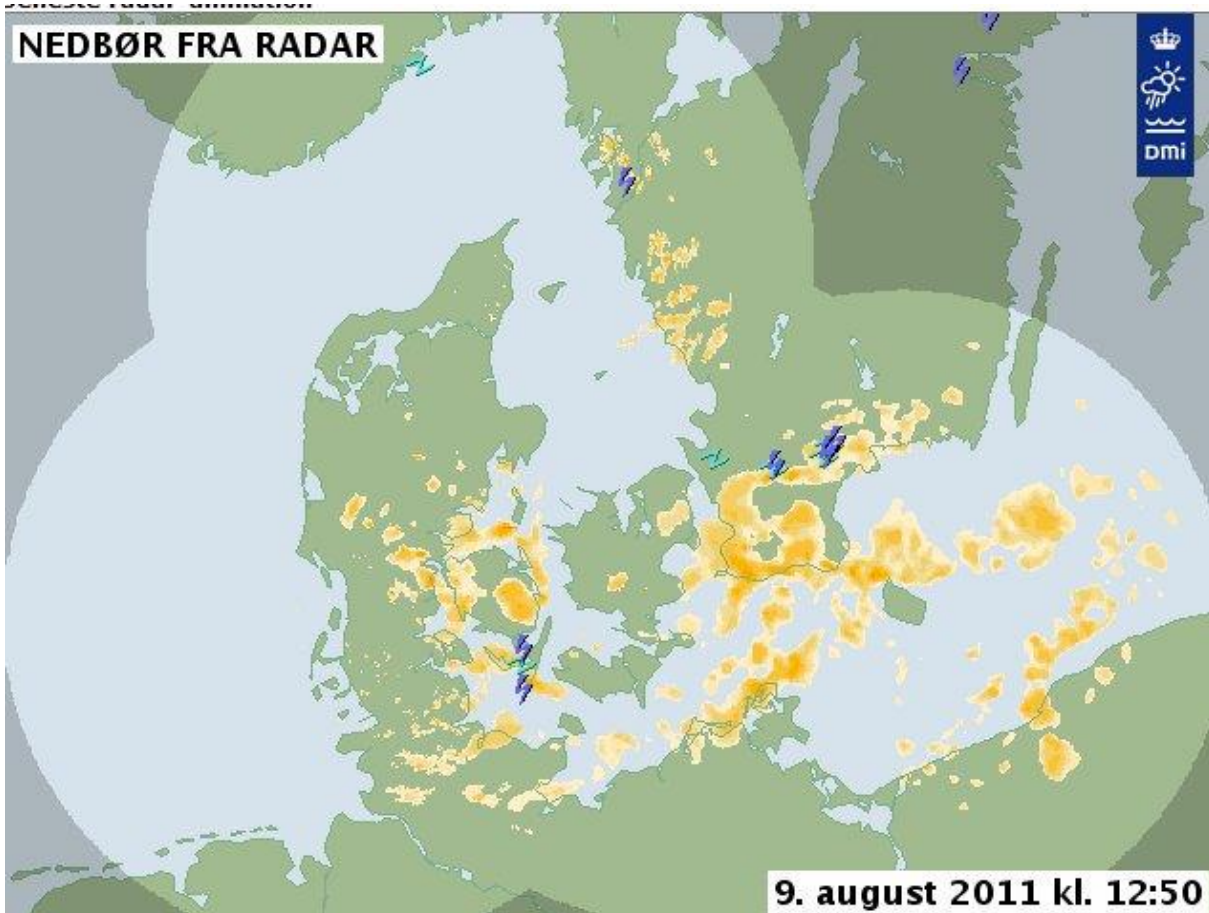


In syntetic speech production, the parameters often are updated every 5 milliseconds.

Chapter 2.	Digital signal processing impulse response, convolution, system function, Fourier, z-transforms	page 7-20
	Matrix description.	page 20-52
	Hints.	page 8-18, 21, 49.
Chapter 3.	Random processing, such as correlation functions, correlation matrices.	
	Random variables	page 58-74
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	Hints.	page 77, 79, 80, 85, 95, 99, 100, 101, 106
Chapter 4.	Signal models, Deterministic and Stochastic approach.	
	Padé, Prony	page 133-154
	Shank	page 154-160
	All-pole Modeling	page 160,165
	Linear prediction	page 165-174
		4.5 not included
	4.6	page 178-188
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Chapter 5.	Levinson-Durbin recursion.	page 215-225, 233-241
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	Hints.	Table 5.1 – 5.4, figure 5.10
Chapter 6.	Lattice FIR and IIR filters, only 6.2 and 6.4.1, 6.4.3	page 289-293, 297, 298, 304-307
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Chapter 7.	Optimal filters. Linear prediction. Wiener filters. Specially FIR filters.	
	FIR- Wiener filter	page 335-345
	IIR- Wiener filter	page 353-371
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Chapter 8.	Spectrum estimation.	
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	8.3 (8.5 see chap 4) , 8.6	page 426-429, 451-472
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Digital Signal Processing application

Radar

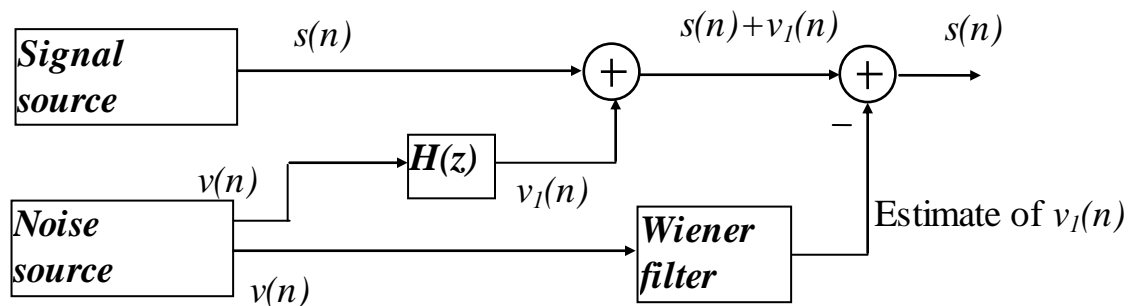


An application from the text book

Noise cancellation (chapter 7, page 349)

A signal is disturbed by additive noise $v_I(n)$.

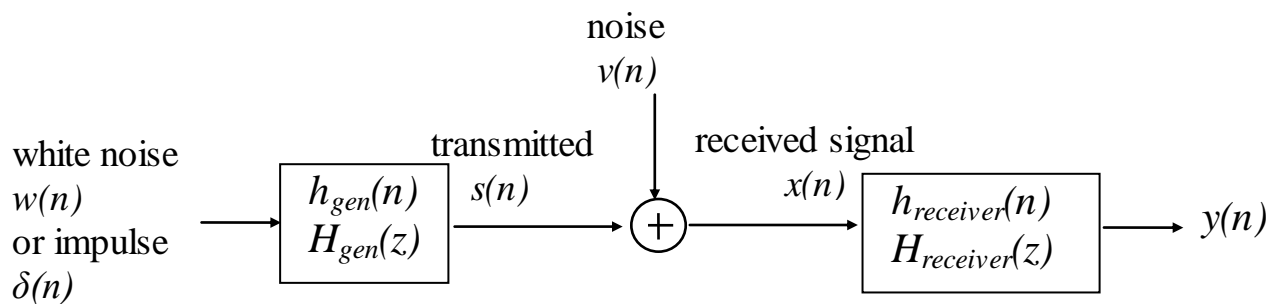
Try to measure the noise $v(n)$ from the source and estimate the noise $v_I(n)$ added to the signal. Then subtract the noise $v_I(n)$ from the received signal.



Optimal signal processing in Hay's book

Chapter 2: **Brief review of digital signal processing.**

Chapter 3: **Brief review of random signals.**



Estimate $H_{gen}(z)$
from properties of
 $s(n)$

Determine
 $H_{receiver}(z)$

The filters $H_{gen}(z)$ and $H_{receiver}(z)$ are of type

FIR

IIR

all-pole IIR

Chapter 4, 5 and 6: **Make a model $H_{gen}(z)$ from the properties of $s(n)$.**

Chapter 7: **Determine $H_{receiver}(z)$.**

Chapter 8: **Estimation of spectra.**

Chapter 2 Digital Signal Processing

Difference equation

$$y(n) = - \sum_{k=1}^p a(k) y(n-k) + \sum_{k=0}^q b(k) x(n-k)$$

MATLAB: `A=[1 0.5 0.5]; B=[1 1]; y=filter(B,A,x);`

Convolution

$$y(n) = \sum_{k=-\infty}^{\infty} h(k) x(n-k)$$

impulse: $\delta(n) = [0 \ 0 \ 0 \ 1 \ 0 \ 0 \ 0]$

unit step: $u(n) = [0 \ 0 \ 0 \ 1 \ 1 \ 1 \dots]$

System function

$$H(z) = \frac{B(z)}{A(z)}$$

Frequency function

$$H(e^{j\omega}) = \frac{B(e^{j\omega})}{A(e^{j\omega})}$$

FIR, IIR filters

FIR: Circuit with impulse response with finite length

Example

$$y(n) = x(n) + x(n-1), \quad h(n) = \delta(n) + \delta(n-1)$$

IIR: Circuit with impulse response with infinite length

Example

$$y(n) = 0.5 y(n-1) + x(n), \quad h(n) = 0.5^n u(n)$$

All-pole IIR-filters

IIR-filters with poles only (all zeroes in origin, $B(z)=\text{constant}$)

Example

$$H(z) = \frac{1}{1 - 0.5 z^{-1}}$$

Method C: Convolution matrix

Use matrix notations

$$x(n) = [\underset{\uparrow}{1} \ 2 \ 3 \ 4], \quad h(n) = [\underset{\uparrow}{4} \ 2 \ 2]$$

$$\begin{bmatrix} x(0) & 0 & 0 \\ x(1) & x(0) & 0 \\ x(2) & x(1) & x(0) \\ x(3) & x(2) & x(1) \\ 0 & x(3) & x(2) \\ 0 & 0 & x(3) \end{bmatrix} \cdot \begin{bmatrix} h(0) \\ h(1) \\ h(2) \end{bmatrix} = \begin{bmatrix} y(0) \\ y(1) \\ y(2) \\ y(3) \\ y(4) \\ y(5) \end{bmatrix}$$

$$\begin{bmatrix} 1 & 0 & 0 \\ 2 & 1 & 0 \\ 3 & 2 & 1 \\ 4 & 3 & 2 \\ 0 & 4 & 3 \\ 0 & 0 & 4 \end{bmatrix} \cdot \begin{bmatrix} 4 \\ 2 \\ 2 \end{bmatrix} = \begin{bmatrix} 4 \\ 10 \\ 18 \\ 26 \\ 14 \\ 8 \end{bmatrix} \quad Xh = y$$

In Matlab: $\mathbf{x}=[1 \ 2 \ 3 \ 4]'$; $\mathbf{X}=\text{convmtx}(\mathbf{x},3)$
 $\mathbf{h}=[4 \ 2 \ 2]'$, $\mathbf{y}=\mathbf{X}*\mathbf{h}$

(In signal processing, all vectors are column vectors)

Properties of matrices

The square matrix A ($n \times n$) is:

symmetrical if $A = A^T$

Hermitian if $A = (A^T)^* = A^H$

invertable if $AA^{-1} = I$

Toeplitz if all diagonals are identical

$$A = \begin{bmatrix} 3 & 4 & 5 \\ 2 & 3 & 4 \\ 1 & 2 & 3 \end{bmatrix}$$

Hermitian (symmetrical) Toeplitz if

$$A = \begin{bmatrix} 3 & 2 & 1 \\ 2 & 3 & 2 \\ 1 & 2 & 3 \end{bmatrix} \quad A = \text{Toep}[3, 2, 1]$$

orthogonal if $A^T A = I$

Linear equation (page 31-34)

*A is a $[n * m]$ matrix*

$Ax = b$ gives

$x = A^{-1}b$ if $n = m, (A \text{ invertable})$

$x = (A^H A)^{-1} A^H b$ if $n > m$

(overdetermined, more equations than variables.) Described more in chapter 4

$x = A^H (A A^H)^{-1} b$ if $n < m$

(underdetermined, less equations than variables)

Eigenvalue:

$$A v = \lambda v, \quad (A - \lambda I) = 0$$

λ eigenvalues, v eigenvectors

$A = V \Lambda V^{-1}$ with eigenvectors in the columns of V , eigenvalues in the diagonal of Λ

Optimisation (minimizing): (page 49)

If z real: $f(z) = z^2$

$$\frac{d}{dz} f(z) = \frac{d}{dz} z^2 = 2z; \quad \frac{d}{dz} z^2 = 0$$

gives $z = 0$ as minimum;

If z is complex: $f(z) = |z|^2 = z^* z$

(z^ is the conjugate of z)*

Derivate with respect to z^* OR z separately while treating the other as a constant.

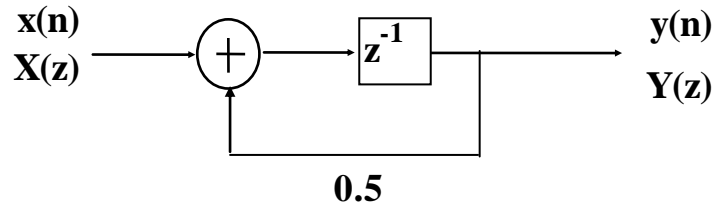
$$\frac{d}{dz} |z|^2 = \frac{d}{dz} z^* z = z^*$$

$$\frac{d}{dz^*} |z|^2 = \frac{d}{dz^*} z^* z = z$$

Setting this derivatives equal to zero gives the same minimum (page 49). This is used sometimes in the textbook.

Example on circuits

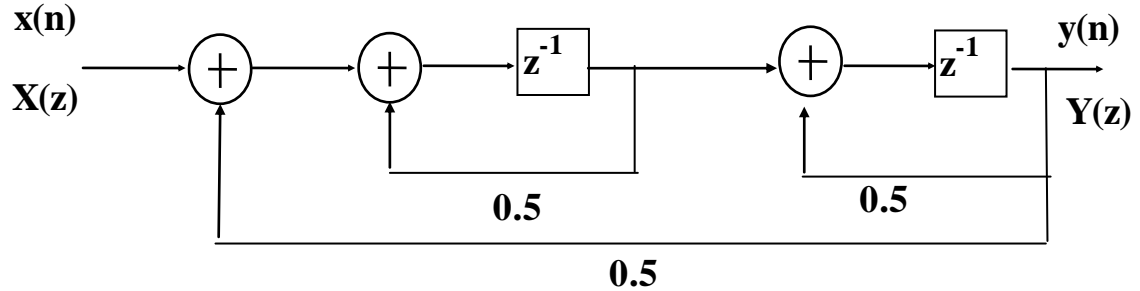
A



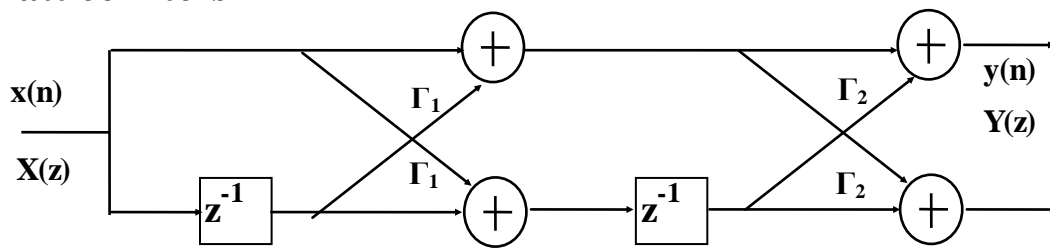
$$y(n) = 0.5 y(n-1) + x(n-1)$$

$$Y(z) = 0.5 z^{-1} Y(z) + z^{-1} X(z)$$

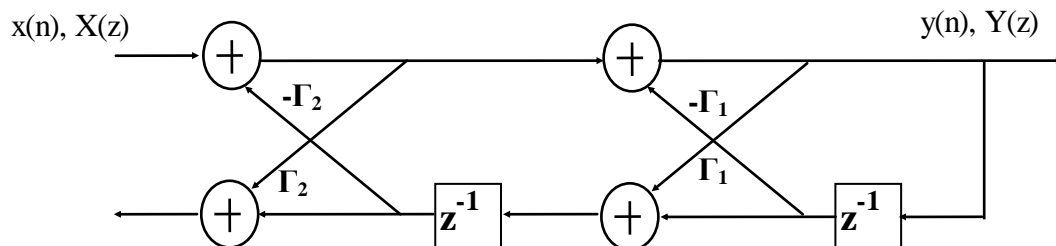
B



C Lattice filters



FIR-lattice filter



IIR-lattice filter

Correlation functions (deterministic)

Autocorrelation function

$$r_x(l) = \sum_{n=-\infty}^{\infty} x(n) x(n-l) \quad (= r_{xx}(l))$$

Cross-correlation function

$$r_{yx}(l) = \sum_{n=-\infty}^{\infty} y(n) x(n-l)$$

$$r_x(l) = x(l) * x(-l)$$

$$r_{yx}(l) = y(l) * x(-l)$$

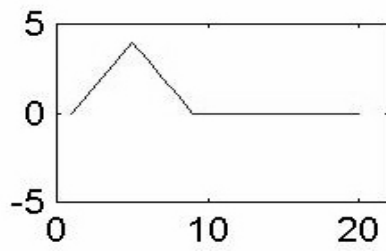
Relation between input and output

$$r_{yx}(l) = h(l) * r_x(l)$$

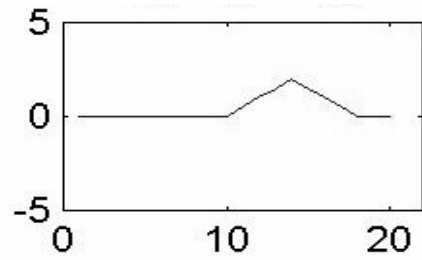
$$r_y(l) = r_h(l) * r_x(l)$$

Example on correlation, echo

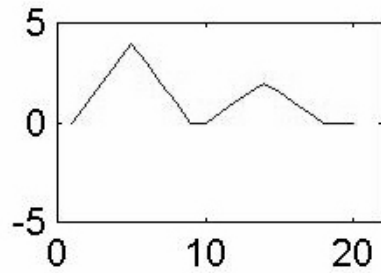
x_1



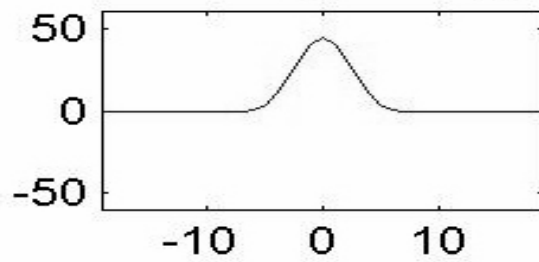
x_2



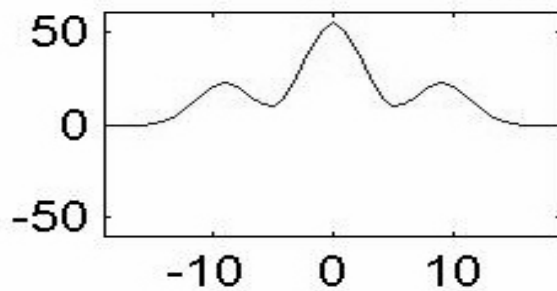
$y=x_1+x_2$



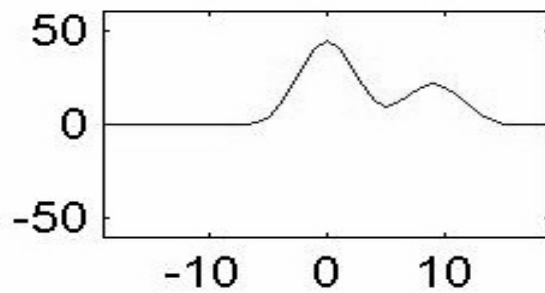
r_{x1}



r_y

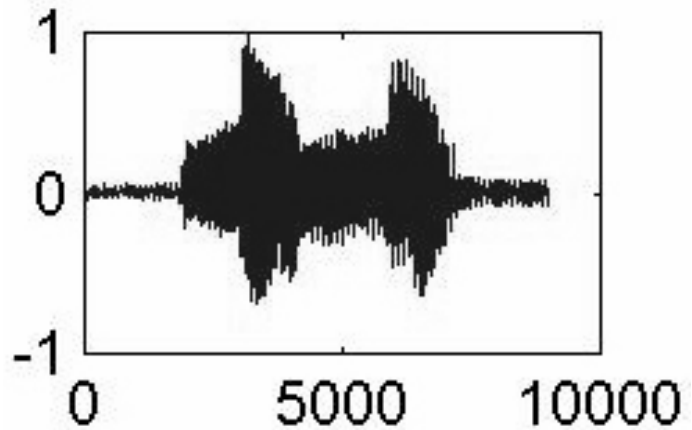


r_{x1y}

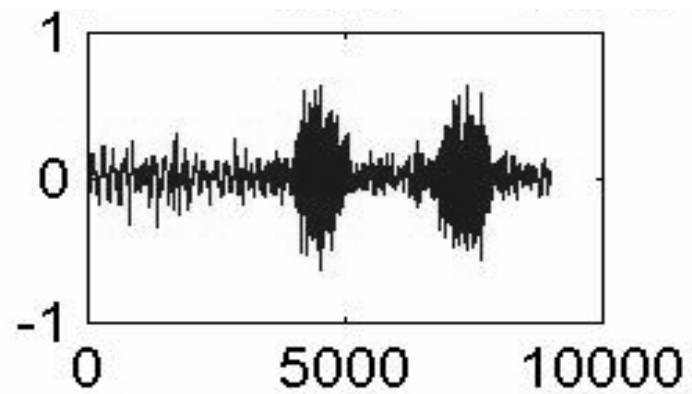


Example of correlation, delay in mobile phones (GSM)

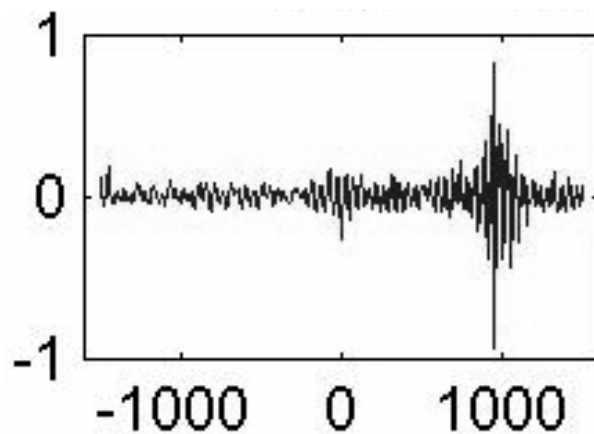
Input signal to the GSM phone



Output signal after GSM



Crosscorrelation



In Matlab: `rx=xcorr(input,output)`

Chapter 3 Discrete-Time Random Processes

Random variables (3.2 page 58-74)

Probability density function $f_X(x)$

Probability distribution function: $F_X(x)$

Expected value (mean): $m = E\{x\} = \int x f_X(x) dx$

Mean-square value: $E\{x^2\} = \int x^2 f_X(x) dx$

Variance:

$$Var[x] = \sigma_x^2 = E\{[x - m]^2\} = \int [x - m]^2 f_X(x) dx$$

General: $y = g(x); \quad E\{y\} = E\{g(x)\} = \int g(x) f_X(x) dx$

Relation:

$$Var[x] = E\{[x - m]^2\} = E\{X^2\} - m^2$$

Correlation. Dependency between random variables x and y

Correlation: $r_{xy} = E\{x y\}$

Covariance: $c_{xy} = E\{[x - m_x][y - m_y]\}$

Stochastic processes (3.3 page 74)

(Wide-sense stationary processes, WSS)

Example A: Sinusoids with random phase

$$x(n) = A \sin(\omega_0 n + \Phi),$$

Φ is a random variable and
 $x(n)$ is a random process.

Example B: Noise (white noise, colored noise).

Example C: Speech signals.

The autocorrelation sequence and the cross-correlation sequence and their Fourier transforms are important in this course.

Autocorrelation sequence:

$$r_x(m) = E\{x(k) x^*(k - m)\}$$

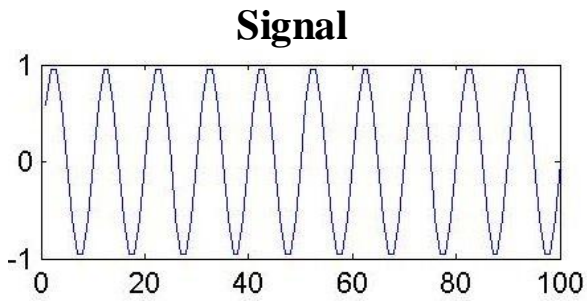
Cross-correlation sequence.

$$r_{xy}(m) = E\{x(k) y^*(k - m)\}$$

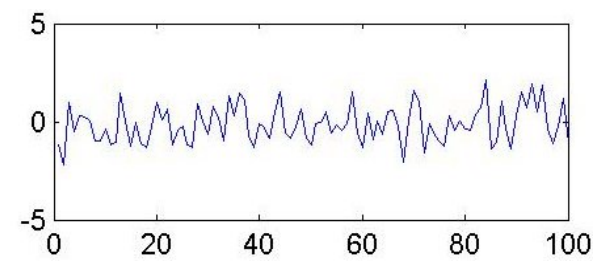
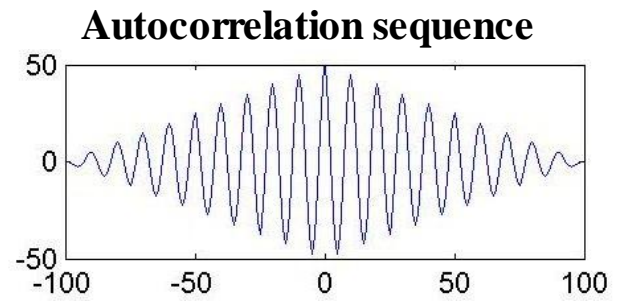
Estimation of the autocorrelation sequence (ergodic processes)

$$r_x(m) = E\{x(k) x(k - m)\} = \frac{1}{N} \sum_{\substack{\text{sum over} \\ N \text{ values}}} x(k) x(k - m)$$

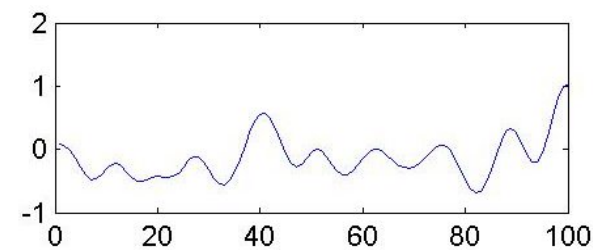
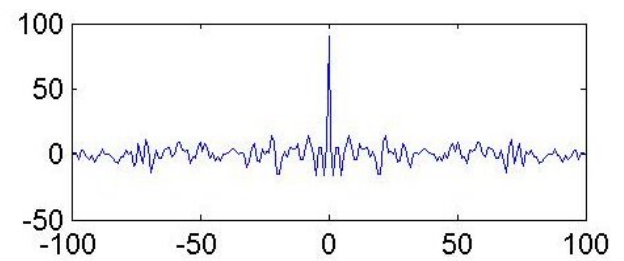
Interpreting of autocorrelation sequence:



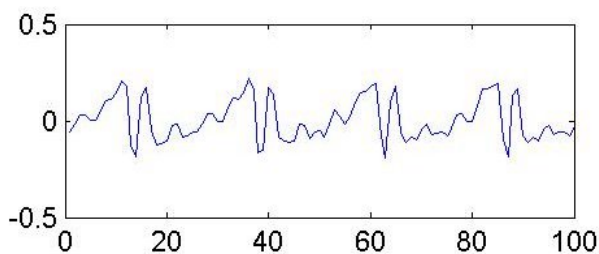
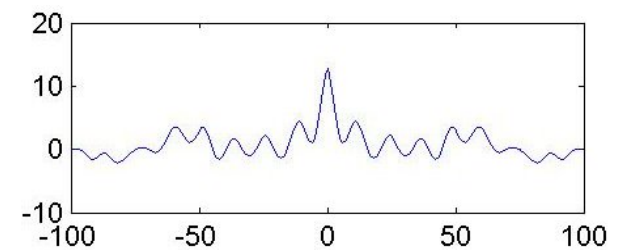
Sinusoid:



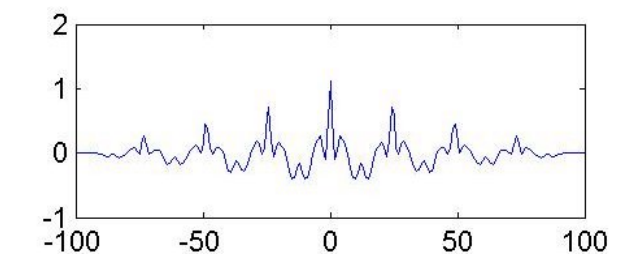
White noise.



Colored noise



Speech signal: Vowel.



Properties of autocorrelation sequence (page 83)

(Wide-sense stationary processes, WSS)

Definition:

$$\begin{aligned} r_x(k) &= E[x(n)x^*(n-k)] = E[x^*(n)x(n-k)]^* = \\ &= E[x(n-k)x^*(n)]^* = E[x(n)x^*(n+k)]^* = r_x(k)^* \end{aligned}$$

Symmetry:

$$r_x(k) = r_x^*(-k)$$

Mean-square value:

$$r_x(0) = E[|x(n)|^2] \geq 0 \quad (\text{positive})$$

Maximum value:

$$r_x(0) \geq |r_x(k)|$$

Non-stationary processes

For signals that are not wide-sense stationary processes, (not WSS), we have to use the definitions (see chapter 4)

$$r_x(k, l) = E\{x(k)x^*(l)\}$$

$$r_{yx}(k, l) = E\{y(k)x^*(l)\}$$

Correlation matrix (WSS)

$$x = [x(0) \ x(1) \dots x(N-1)]^T$$

$$R_x = E[x x^H] =$$

$$= \begin{bmatrix} r_x(0) & r_x^*(1) & r_x^*(2) & \dots & r_x^*(p) \\ r_x(1) & r_x(0) & r_x^*(1) & \dots & r_x^*(p-1) \\ r_x(2) & r_x(1) & r_x(0) & \dots & r_x^*(p-2) \\ & & \cdot & & \\ r_x(p) & r_x(p-1) & r_x(p-2) & \dots & r_x(0) \end{bmatrix}$$

Properties of the correlation matrix

Hermitian Toeplitz

Toeplitz if real-valued process

Eigenvalues are real and non-negative

Estimate of the correlation function

$$\hat{r}_x(k) = \frac{1}{N} \sum_{n=0}^{N-1} x(n) x^*(n-k)$$

Estimate of the cross-correlation function

$$\hat{r}_{xy}(k) = \frac{1}{N} \sum_{n=0}^{N-1} x(n) y^*(n-k)$$

Power spectrum of random process (3.3.8 page 94):
(Wide-sense stationary processes, WSS)

$x(n)$ is a wide sense stationary random process
(WSS, $x(n)$ real-valued, $h(n)$ real) with
autocorrelation $r_x(k)$

The Fourier transform and the z-transform are given
by:

The Fourier transform of $r_x(k)$:

$$P_x(e^{j\omega}) = \sum r_x(k) e^{-j\omega k}$$

The Z-transform of $r_x(k)$:

$$P_x(z) = \sum r_x(k) z^{-k}$$

Properties

Symmetry (real processes)

:

$$P_x(e^{j\omega}) = P_x(e^{-j\omega})$$

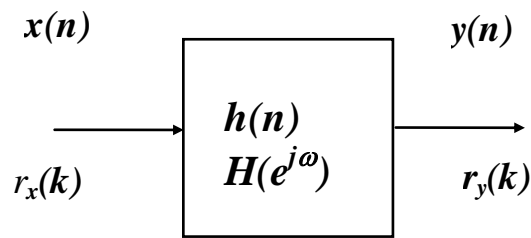
Positive:

$$P_x(e^{j\omega}) \geq 0$$

Total power:

$$r_x(0) = \frac{1}{2\pi} \int_{-\pi}^{\pi} P_x(e^{j\omega}) d\omega$$

**Filtering of random processes,
(3.4 page 99, 100, 101):**



Input-output relation

$$y(n) = x(n) * h(n) = \sum_{k=-\infty}^{\infty} x(k) h(n - k)$$

Autocorrelation function for the output

$$r_y(k) = E\{y(n) y(n - k)\} = \sum_{l=-\infty}^{\infty} \sum_{m=-\infty}^{\infty} h(l) r_x(m - l + k) h(m)$$

Cross correlation functions

$$r_{yx}(k) = E\{y(n) x(n - k)\} = \sum_{l=-\infty}^{\infty} h(l) r_x(k - l)$$

$$r_{xy}(k) = E\{x(n) y(n - k)\} = \sum_{l=-\infty}^{\infty} h(l) r_x(k + l)$$

Using convolution and power spectra

Define $r_h(k) = \sum h(l)h(l+k) = h(k) * h(-k)$

Correlation functions

$$r_y(k) = r_x(k) * h(k) * h(-k) = r_x(k) * r_h(k)$$

$$r_{yx}(k) = r_x(k) * h(k)$$

$$r_{xy}(k) = r_x(k) * h(-k)$$

Spectra

$$P_y(e^{j\omega}) = P_x(e^{j\omega}) |H(e^{j\omega})|^2$$

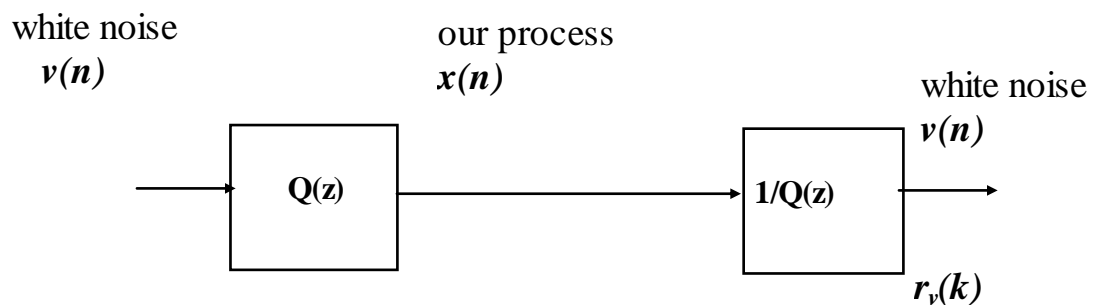
$$P_{yx}(e^{j\omega}) = P_x(e^{j\omega}) H(e^{j\omega})$$

$$P_{xy}(e^{j\omega}) = P_x(e^{j\omega}) H^*(e^{j\omega})$$

$$P_y(z) = P_x(z) H(z) H\left(\frac{1}{z}\right)$$

Spectral factorization (3.5 page 104)

$x(n)$ is a WSS process with autocorrelation $r_x(k)$. We assume that the process are generated from white noise $v(n)$ filtered in a filter with system function $Q(z)$, Then, $v(n)$ is called the innovation process of the process $x(n)$.



$$r_v(k) = \sigma_0^2 \delta(k) \quad r_x(k)$$

$$P_v(z) = \sigma_0^2 \quad P_x(z) = \sigma_0^2 Q(z) Q^*(1/z^*)$$

Can we find the filter $Q(z)$ from $x(n)$ and $r_x(k)$?

Is $Q(z)$ stable and causal?

Is $1/Q(z)$ stable and causal?