

Discrete-time Systems in the Time Domain

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Discrete-time Systems

- “A **discrete-time system** is an entity that processes a discrete-time input signal $x(k)$ to produce a discrete-time output signal $y(k)$.”
- **Digital signals** are in both amplitude and time.
- A system associated with digital signals is a **digital filter**.
- “A finite-dimensional linear time-invariant (LTI) discrete-time system can be represented in the time domain by a constant-coefficient difference equation”

$$y(k) + \sum_{i=1}^N a_i y(k-i) = \sum_{i=0}^M b_i x(k-i), \quad \text{i.e., } a_0 = 1$$

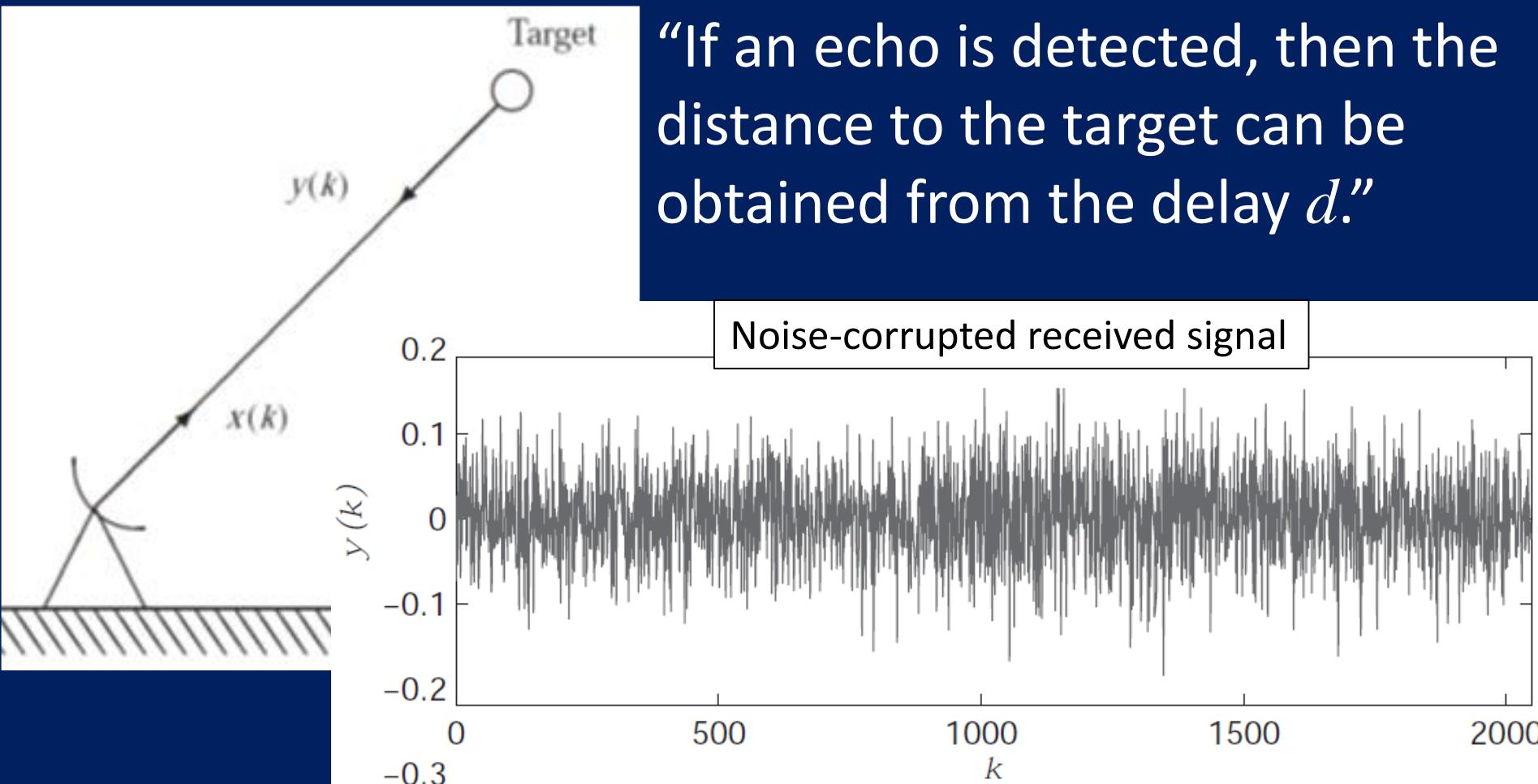
- “Many problems can be modeled as discrete-time systems using difference equations.”

Example: Radar

- “A radar antenna transmits an EM wave $x(k)$ into space. When a target is illuminated by the radar, some of signal energy is reflected back and returns to the radar receiver.”
- “The received signal $y(k)$ can be modeled using the difference equation”
$$y(k) = ax(k - d) + \eta(k)$$
- “The first term represents the echo of the transmitted signal with delay d proportional to the time of flight.”
- “The second term accounts for random noise picked up and amplified by the receiver.”
- “Typically, the echo is very faint, i.e., $0 < a \ll 1$.”

Example: Radar

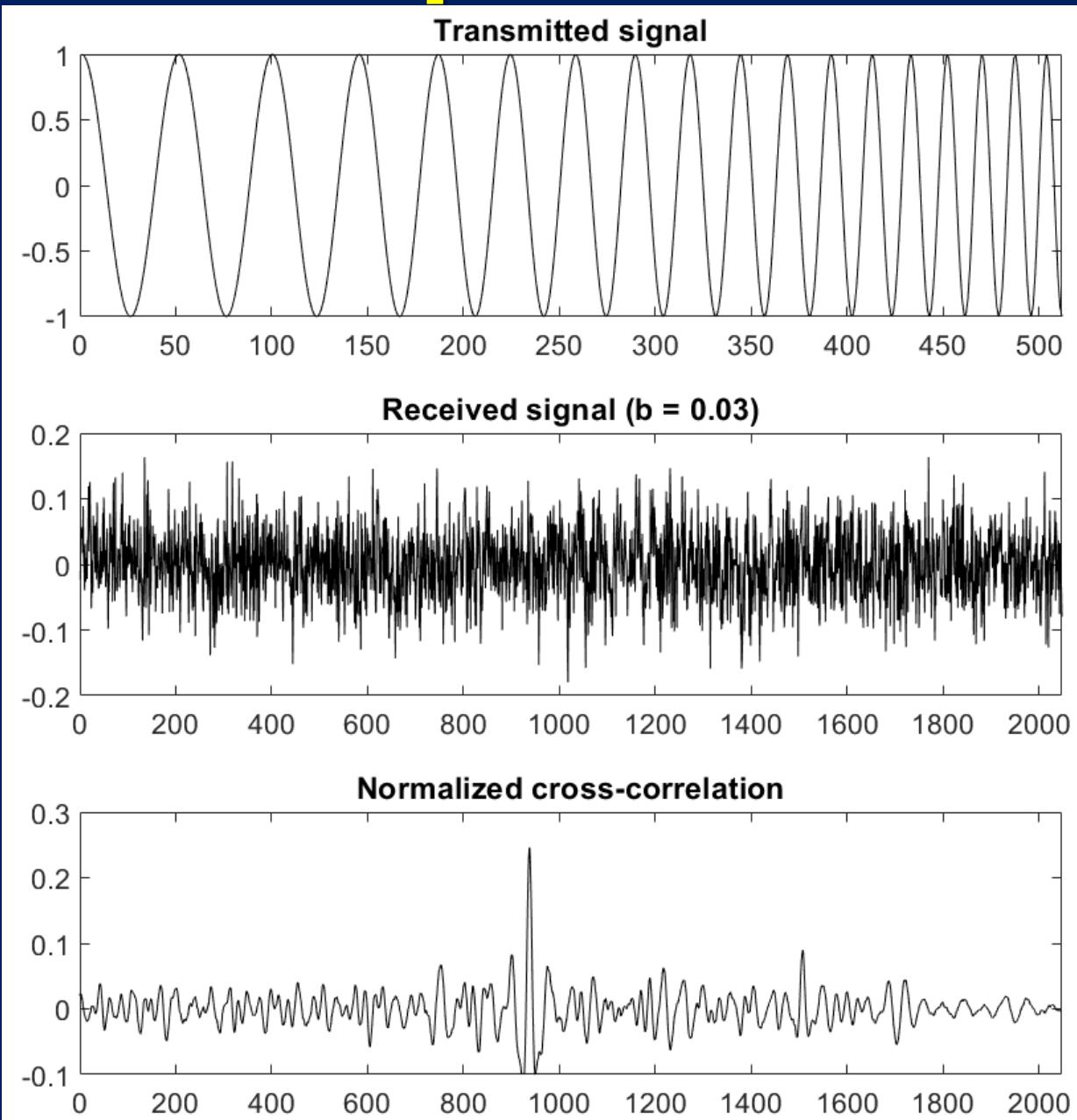
“The objective in processing the received signal is to determine whether or not an echo is present.”



Example: Radar

- “If T is the sampling interval, then the time of flight in seconds is $\tau = dT$.”
- The distance to the target is $r = cdT/2$ where c is the speed of light and the factor 2 arises because the time of flight is a two-way travel time.
- “To detect if an echo is present in the received signal, we compute the normalized cross-correlation which measures the degree to which the received signal $y(k)$ is similar to the transmitted signal $x(k)$.”

Example: Radar



Classification of Discrete-time Signals

- **Finite signal:** signal is nonzero for a finite number of samples $x(k) = 0, k \notin [N_1, N_2]$ where $N_1 \leq N_2$. Otherwise, it is called *infinite signal*.
- **Causal signal:** $x(k) = 0, k < 0$. Otherwise, it is called *noncausal signal*.
- **Periodic signal:** $x(k + N) = x(k)$ where N is the period.
- **Bounded signal:** $|x(k)| \leq B_x, B_x > 0$. Otherwise, it is an *unbounded signal*.

Norm of Signals

- Signals can be thought of as vectors.
- The length of vectors can be measured by

- l_1 norm $\|x\|_1 \stackrel{\text{def}}{=} \sum_{k=-\infty}^{\infty} |x(k)|$

- l_2 norm $\|x\|_2 \stackrel{\text{def}}{=} \left[\sum_{k=-\infty}^{\infty} |x(k)|^2 \right]^{1/2}$

- **Absolutely summable signal:** $\|x\|_1 < \infty$
- **Square summable signal:** $\|x\|_2 < \infty$

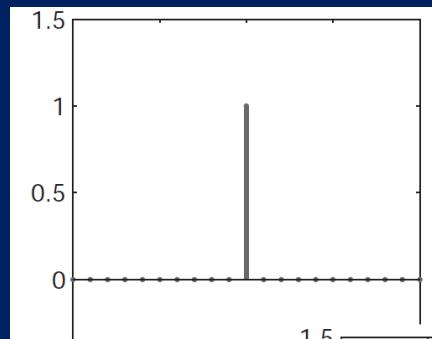
Energy and Power Signals

- **Energy** of signal is defined as $E_x = \sum_{k=-\infty}^{\infty} |x(k)|^2$
- Since $E_x = \|x\|_2^2$, the energy is finite if $x(k)$ is square summable.
- **Energy signal**: a signal with finite energy $E_x < \infty$
- **Instantaneous power**: $p(x) = |x(k)|^2$
- **Average power**: $P_x = \lim_{N \rightarrow \infty} \frac{1}{2N+1} \sum_{k=-N}^N |x(k)|^2$
- **Power signal**: a finite signal with nonzero average power

Common Signals

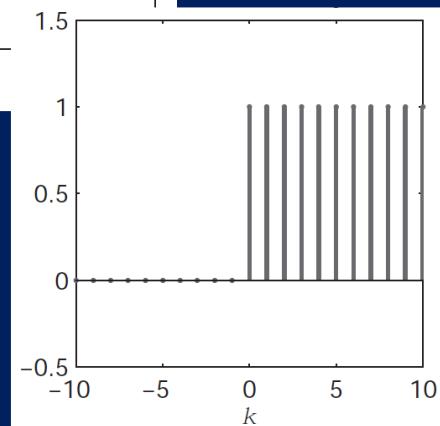
- **Unit impulse:**

$$\delta(k) \stackrel{\text{def}}{=} \begin{cases} 1, & k = 0 \\ 0, & k \neq 0 \end{cases}$$

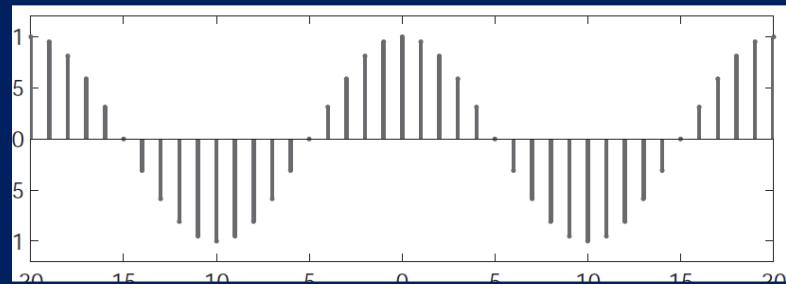
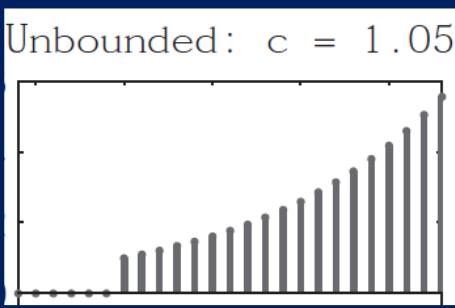
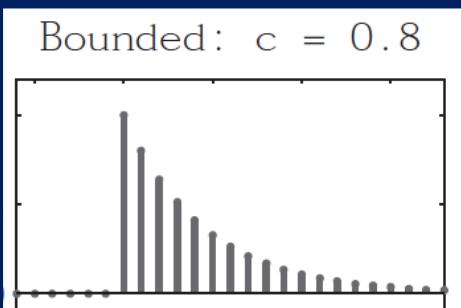


- **Unit step:**

$$\mu(k) \stackrel{\text{def}}{=} \begin{cases} 1, & k \geq 0 \\ 0, & k < 0 \end{cases}$$



- **Causal exponential:** $x(k) = c^n \mu(k)$



- **Periodic signal:** $x(k) = \cos(2\pi fkT)$

Discrete-time Systems

- **Linear system:** S is a linear system if

$$x(k) = ax_1(k) + bx_2(k) \Rightarrow y(k) = ay_1(k) + by_2(k)$$

Otherwise, the system S is nonlinear.

- **Time-invariant system:** S is a time-invariant system if the output produced by the shifted input $x(k - m)$ is $y(k - m)$. Otherwise, the system S is time-varying.

- **Causal system:** “For a physical system operating in *real time*, the output at the present time k cannot depend on the future input $x(m)$, $m > k$, because the input has not yet occurred.”

- “A discrete-time system is causal if for each time k ”
 $x_1(m) = x_2(m)$, for $m \leq k \Rightarrow y_1(k) = y_2(k)$

Discrete-time Systems

- “When signal processing is not performed in real time, noncausal systems can be used to process the data *offline* in batch mode, where future samples of the input are available.”
- **Stable system:** S is a bounded-input bounded-output (BIBO) stable if every bounded input $x(k)$ produces a bounded output $y(k)$:

$$|x(k)| \leq B_x \Rightarrow |y(k)| \leq B_y$$

- **Passive system:** energy does not increase: $E_y \leq E_x$
- **Lossless system:** energy stays the same: $E_y = E_x$
- “Lossless *physical* systems contain energy storage elements (spring, mass, capacitor, inductor) without energy dissipative elements (resistor, damper)

Difference Equations

- “The output of a causal LTI system S at time k can be expressed as”

$$y(k) = \sum_{i=0}^M b_i x(k-i) - \sum_{i=1}^N a_i y(k-i)$$

- “When causal inputs are used, the output or response of a discrete-time system depends on both the input $x(k)$ and the initial condition represented by a vector $y_0 \in \mathbb{R}^N$ of past outputs,”

$$y_0 \stackrel{\text{def}}{=} [y(-1), y(-2), \dots, y(-N)]^T$$

- “For a linear system, the contributions to the output from initial condition y_0 and input $x(k)$ can be considered separately.”

Zero-input Response

- Zero-input response of a discrete-time system S is the solution of the system

$$y(k) + \sum_{i=1}^N a_i y(k-i) = 0, \quad y_0 \neq 0$$

when the input is $x(k) = 0$ and is denoted as $y_{zi}(k)$.

- To solve the system for a zero-input response, let's use the trial solution of the form $y(k) = z^k$ where z is a complex scalar to be determined.
- Substituting the trial solution into the system and multiplying both sides by z^{N-k} yields the **characteristic polynomial**,

$$a(z) = z^N + a_1 z^{N-1} + \cdots + a_N = 0$$

Zero-input Response: Simple Mode

- The factored form of the characteristic polynomial is
$$a(z) = (z - p_1)(z - p_2) \cdots (z - p_N) = 0$$
where p_i are the roots. So, z must be equal to the roots.
- If the characteristic polynomial has N distinct roots, also called simple roots, the general solution is the linear combination

$$y_{zi}(k) = \sum_{i=1}^N c_i p_i^k, \quad k \geq -N$$

where $c_i p_i^k$ are called the **simple natural modes** of the system.

- Applying the initial conditions yields a linear system for determining the value of the coefficients c_i .

Example

Consider the 2D discrete-time system

$$y(k) - 0.6y(k-1) + 0.05y(k-2) = 2x(k) + x(k-1)$$

with initial condition $y(-1) = 3$ and $y(-2) = 2$.

The characteristic polynomial of the system is

$$a(z) = z^2 - 0.6z + 0.05 = (z - 0.5)(z - 0.1)$$

and has simple roots $p_1 = 0.5, p_2 = 0.1$

The zero-input response is then of the form

$$y_{zi}(k) = c_1(0.5)^k + c_2(0.1)^k$$

Applying the initial conditions $y_{zi}(-1) = 3, y_{zi}(-2) = 2$ yields the linear system

$$\begin{bmatrix} 2 & 10 \\ 4 & 100 \end{bmatrix} \begin{bmatrix} c_1 \\ c_2 \end{bmatrix} = \begin{bmatrix} 3 \\ 2 \end{bmatrix} \rightarrow c_1 = 1.75, c_2 = -0.05$$

Zero-input Response: Multiple Mode

When root p occurs r times, p is referred to as a root of multiplicity r , and generates a **multiple natural mode** of the form $(c_1 + c_2k + \cdots + c_rk^{r-1})p^k$

Example: System $y(k) + y(k-1) + 0.25y(k-2) = 3x(k)$ with initial condition $y(-1) = -1$ and $y(-2) = 6$ has the corresponding characteristic polynomial as

$$a(z) = z^2 + z + 0.25 = (z + 0.5)^2$$

whose root is $p_1 = -0.5$ of multiplicity 2.

The zero-input response is then of the form

$$y_{zi}(k) = (c_1 + c_2k)(-0.5)^k$$

Applying the initial conditions yields the linear system

$$\begin{bmatrix} -2 & 2 \\ 4 & -8 \end{bmatrix} \begin{bmatrix} c_1 \\ c_2 \end{bmatrix} = \begin{bmatrix} -1 \\ 6 \end{bmatrix} \rightarrow c_1 = -0.5, c_2 = -1$$

Zero-input Response: Complex Mode

When roots are complex, they always occur in conjugate pairs since the coefficients of $a(z)$ are real. Complex conjugate roots $p_{1,2} = r \exp(\pm j\theta)$ form the **complex mode** of the form $r^k [c_1 \cos(k\theta) + c_2 \sin(k\theta)]$

Hint: use Euler identity $\exp(\pm j\theta) = \cos \theta \pm j \sin \theta$

Example: System $y(k) + 0.49y(k - 2) = 3x(k)$ with initial condition $y(-1) = 4$ and $y(-2) = 2$ has the corresponding characteristic polynomial as

$$a(z) = z^2 + 1 = (z - j0.7)(z + j0.7)$$

The zero-input response is then of the form

$$y_{zi}(k) = (0.7)^k [c_1 \cos(\pi k/2) + c_2 \sin(\pi k/2)], \quad \theta = \pi/2$$

Applying the initial conditions yields $c_1 = 0.98, c_2 = -2.8$

Zero-state Response

Zero-state response of a LTI discrete-time system is the output corresponding to an arbitrary input $x(k)$ when the initial condition vector is zero, i.e.,

$$y(k) + \sum_{i=1}^N a_i y(k-i) = \sum_{i=0}^M b_i x(k-i), \quad y_0 = 0$$

Consider a special case of **causal exponential input** of the form

$$x(k) = A p_0^k \mu(k)$$

where A is amplitude and p_0 exponential factor.

If $M \leq N$, then the input polynomial associated with the coefficients of the input is

$$b(z) = b_0 z^N + b_1 z^{N-1} + \cdots + b_M z^{N-M}$$

Zero-state Response

Suppose that the input is a causal exponential and that the characteristic polynomial $a(z)$ has $N+1$ distinct roots. The zero-state response has a form similar to the zero-input response:

$$y_{zs}(k) = \sum_{i=1}^N d_i p_i^k \mu(k)$$

“The weighting coefficient $d \in \mathbb{R}^{N+1}$ can be computed from the polynomials $a(z)$ and $b(z)$ as

$$d_i = \left. \frac{A(z - p_i)b(z)}{(z - p_0)a(z)} \right|_{z=p_i}, \quad 0 \leq i \leq N$$

where p_i are the roots of the denominator.”

Example

Consider the system

$$y(k) - 0.6y(k-1) + 0.05y(k-2) = 2x(k) + x(k-1)$$

with the input $x(k) = 0.8^{k+1} \mu(k)$, i.e., $A = 0.8$, $p_0 = 0.8$

“The characteristic polynomial of this system is”

$$a(z) = z^2 - 0.6z + 0.05 = (z - 0.5)(z - 0.1)$$

“The input polynomial of this system is”

$$b(z) = 2z^2 + z = 2z(z + 0.5)$$

We then have

$$\frac{Ab(z)}{(z - p_0)a(z)} = \frac{1.6z(z + 0.5)}{(z - 0.8)(z - 0.5)(z - 0.1)}$$

Example (continued)

The weighting coefficients are

$$d_0 = \frac{1.6(0.8)(1.3)}{0.3(0.7)} = 7.92, \quad p_0 = 0.8$$

$$d_1 = \frac{1.6(0.5)(1.0)}{-0.3(0.4)} = -6.67, \quad p_1 = 0.5$$

$$d_2 = \frac{1.6(0.1)(0.6)}{-0.7(-0.4)} = 0.343, \quad p_2 = 0.1$$

The zero-state response is then

$$y_{zs}(k) = [7.92(0.8)^k - 6.67(0.5)^k + 0.343(0.1)^k] \mu(k)$$

The complete response is $y(k) = y_{zi}(k) + y_{zs}(k)$

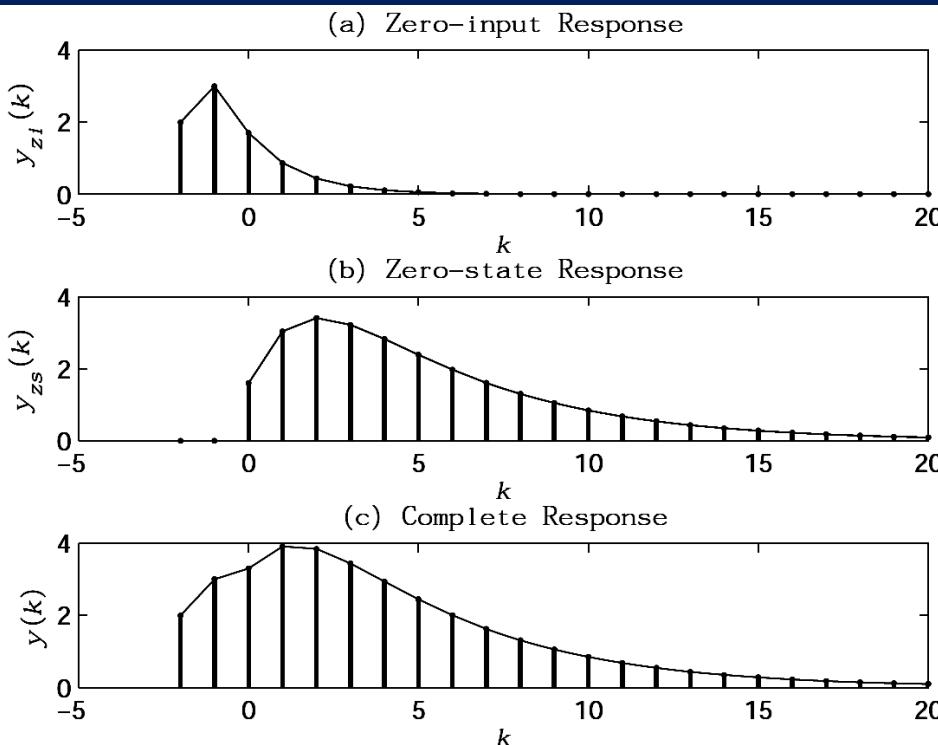
Example (continued)

Suppose the initial condition is $y_0 = [3, 2]$, that is $y(-1) = 3, y(-2) = 2$. The zero-input response is

$$y_{zi}(k) = 1.75(0.5)^k - 0.05(0.1)^k$$

So, the complete response of the system is

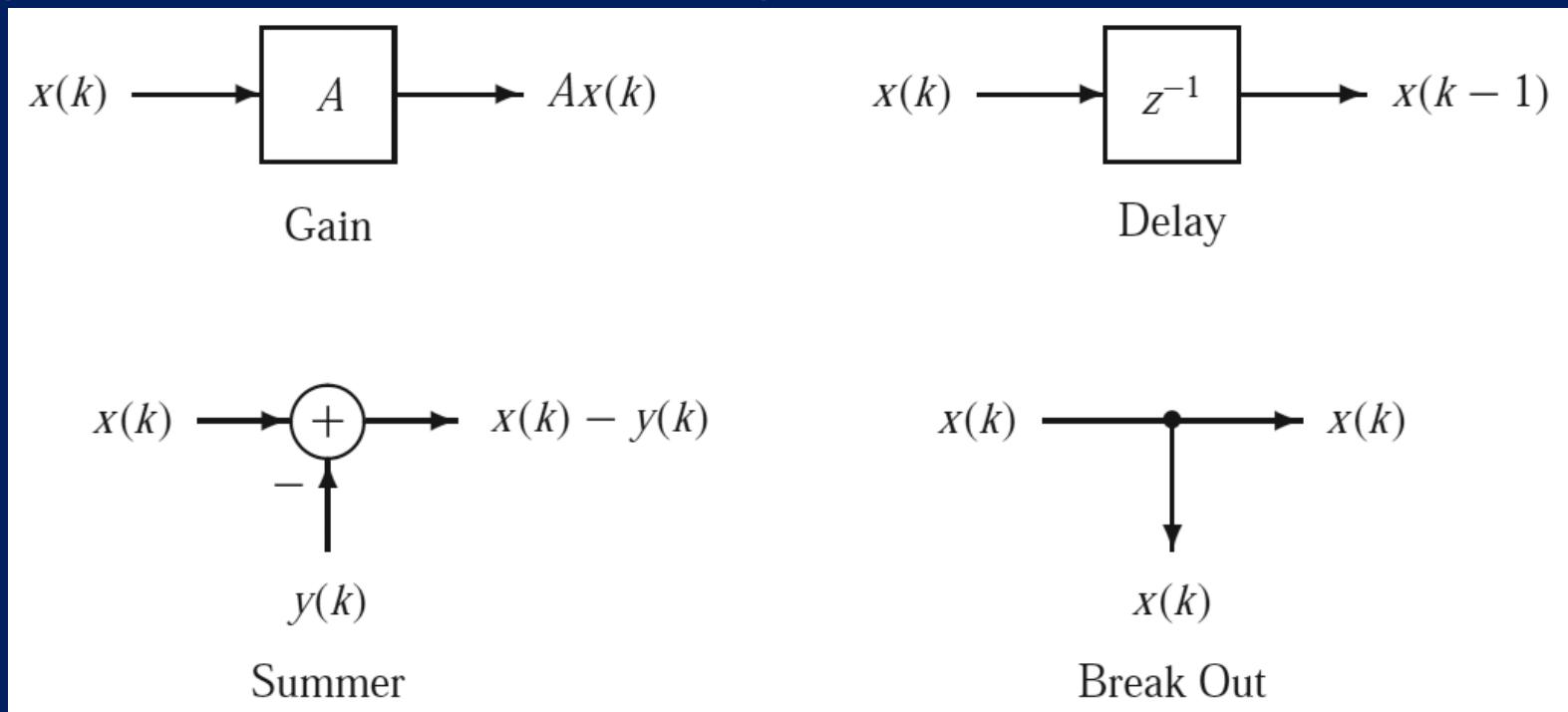
$$y(k) = 1.75(.5)^k - .05(.1)^k + \left[7.92(.8)^k - 6.67(.5)^k + .343(.1)^k \right] \mu(k)$$



“Note that for $n \gg 1$, the complete response is dominated by the zero-state response because the zero-input response quickly dies out.”

Block Diagram

- “Discrete-time systems can be displayed graphically in the form of block diagrams.”
- “A block diagram is a set of blocks that represent processing units interconnected by directed line segments that represent signals.”

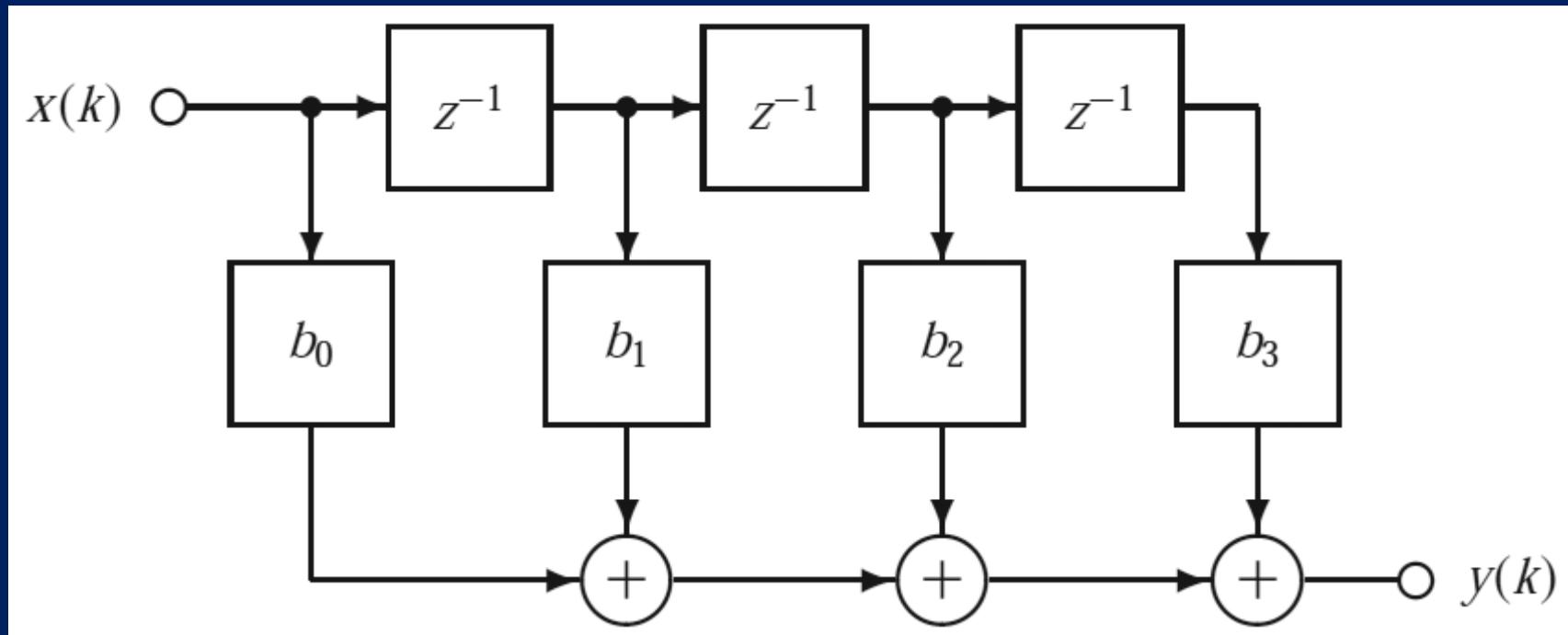


Block Diagram: Example 1

Consider a **moving average filter** whose output is a weighted sum of the past inputs:

$$y(k) = \sum_{i=0}^M b_i x(k-i)$$

When $M = 3$, the block diagram of moving average filter is



Block Diagram: Example 2

A LTI system $y(k) = \sum_{i=0}^M b_i x(k-i) - \sum_{i=1}^N a_i y(k-i)$

can be rewritten as

$$y(k) = b_0 x(k) + \sum_{i=1}^D [b_i x(k-i) - a_i y(k-i)]$$

by zero padding the coefficient vectors a and b so that $a, b \in \mathbb{R}^{D+1}$ where $D = \max\{N, M\}$ is the system dimension.

When $D = 2$, using intermediate signals, the output can be defined recursively by

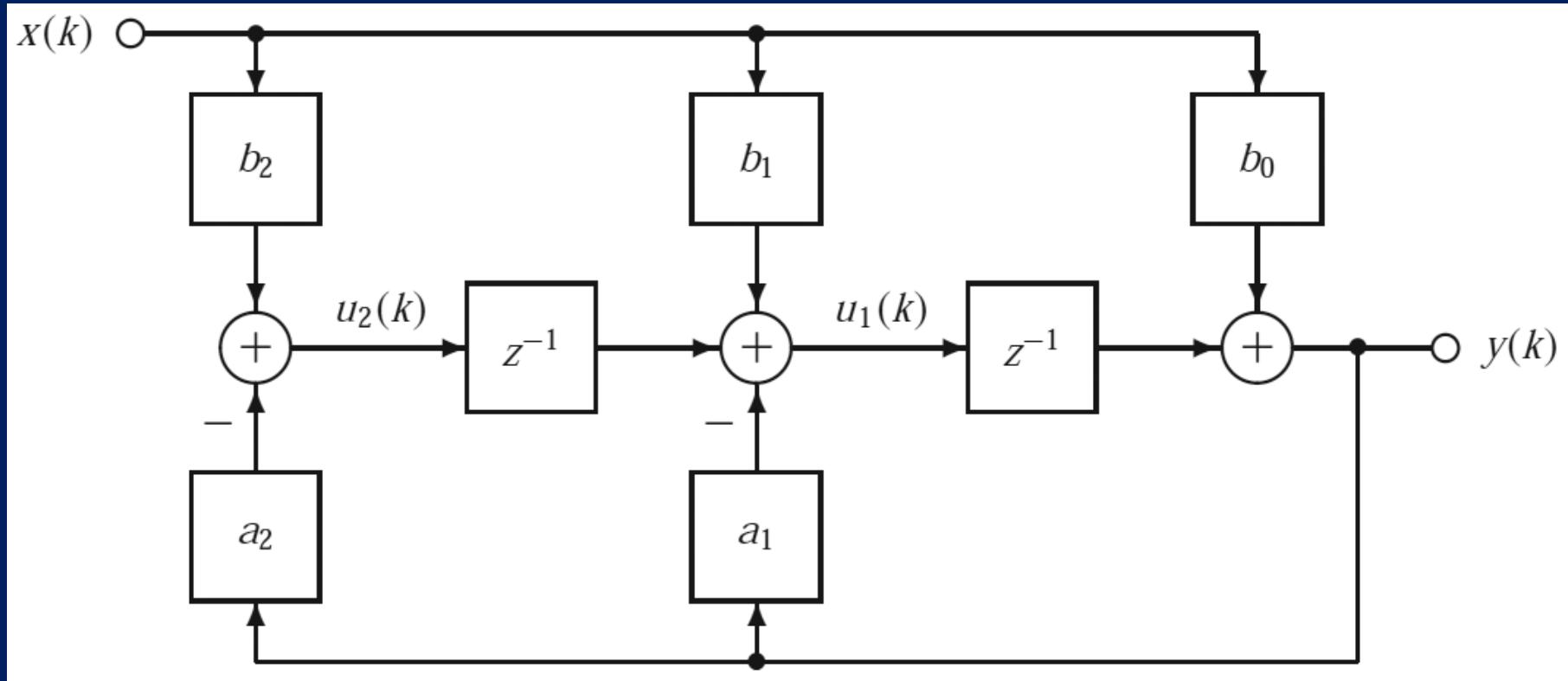
$$u_2(k) = b_2 x(k) - a_2 y(k)$$

$$u_1(k) = b_1 x(k) - a_1 y(k) + u_2(k-1)$$

$$y(k) = b_0 x(k) + u_1(k-1)$$

Block Diagram: Example 2

When $D = 2$, the block diagram for the system is



$$u_2(k) = b_2 x(k) - a_2 y(k)$$

$$u_1(k) = b_1 x(k) - a_1 y(k) + u_2(k-1)$$

$$y(k) = b_0 x(k) + u_1(k-1)$$

Block Diagram: Example 2

For an arbitrary value of D , the output can then be defined recursively as

$$u_D(k) = b_D x(k) - a_D y(k)$$

$$u_{D-1}(k) = b_{D-1} x(k) - a_{D-1} y(k) + u_D(k-1)$$

⋮

$$u_1(k) = b_1 x(k) - a_1 y(k) + u_2(k-1)$$

$$y(k) = b_0 x(k) + u_1(k-1)$$

Impulse Response

- “The **impulse response** of a LTI system is the zero-state response $h(k)$ produced by the unit impulse input.”

$$x(k) = \delta(k) \Rightarrow h(k) = y_{zs}(k)$$

- The impulse response can be used to compute the zero-state response of any input.

Consider a system $y(k) = \sum_{i=0}^M b_i x(k-i)$

The impulse response of this system is

$$h(k) = \sum_{i=0}^M b_i \delta(k-i)$$

Impulse Response

Recall the property of unit impulse

$$\delta(k - i) = \begin{cases} 1, & k = i \\ 0, & k \neq i \end{cases}$$

Consequently, the impulse response is

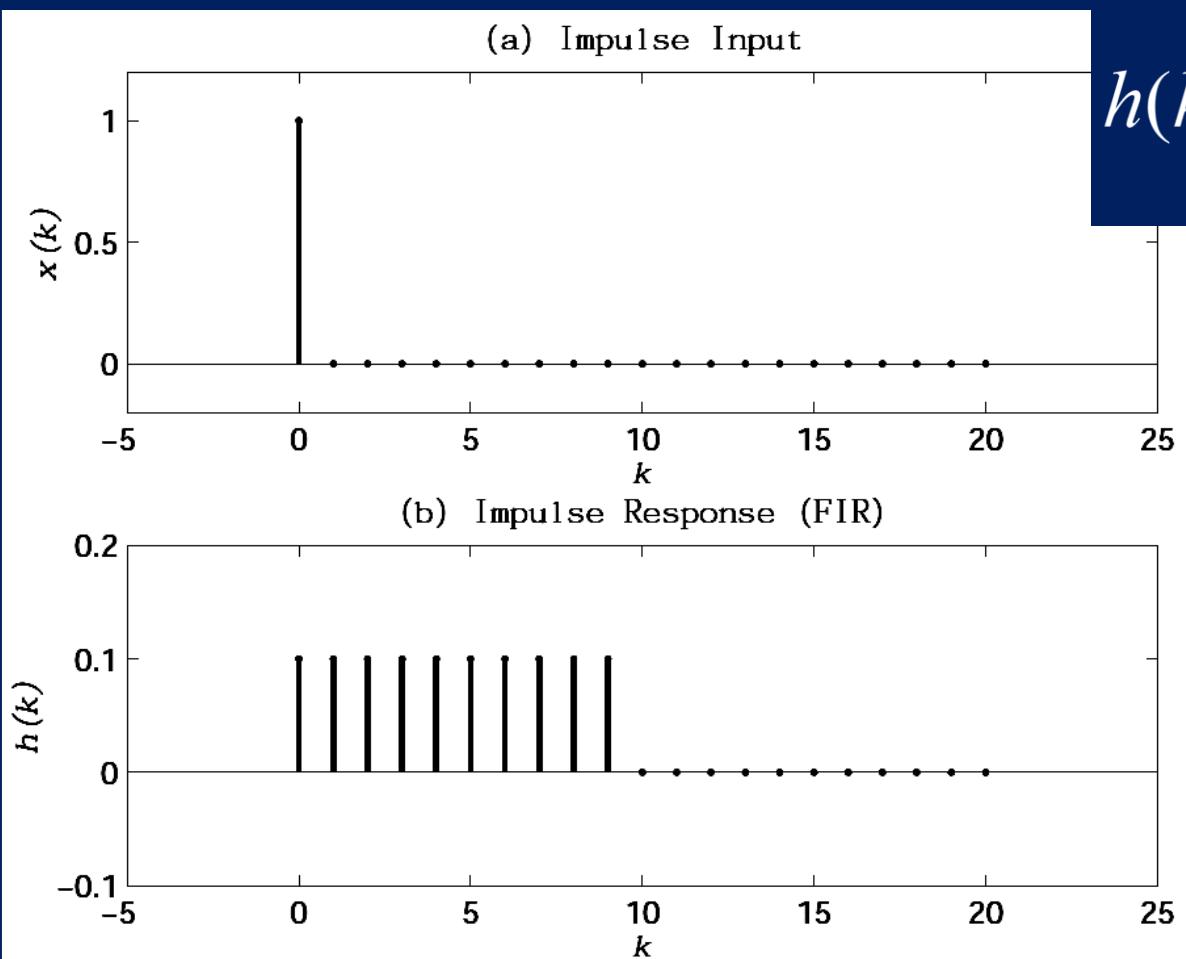
$$h(k) = \begin{cases} b_k, & 0 \leq k \leq M \\ 0, & M < k \leq \infty \end{cases}$$

“A linear system whose impulse response contains a finite number of nonzero samples is called a **finite impulse response (FIR)** system. Otherwise, the system is an **infinite impulse response (IIR)** system.”

FIR System: Example

Consider a moving average filter $y(k) = \frac{1}{M} \sum_{i=0}^{M-1} x(k-i)$

The impulse response of this filter is



$$h(k) = \begin{cases} 1/M, & 0 \leq k \leq M \\ 0, & M < k \leq \infty \end{cases}$$

$$M = 10$$

Schilling and Harris (2012, p.97-98)

IIR System

Consider a LTI system

$$y(k) = \sum_{i=0}^M b_i x(k-i) - \sum_{i=1}^N a_i y(k-i), \quad N \geq 1$$

A systematic technique for solving this general system is the Z transform which will be introduced later.

“Suppose that $M \leq N$ and the roots of the characteristic polynomial $a(z)$ are simple and nonzero.”

The impulse response in this case is

$$h(k) = c_0 \delta(k) + \sum_{i=1}^N c_i (p_i)^k \mu(k)$$

where p_i are the roots of $a(z)$.

IIR System

The coefficient vector $c \in \mathbb{R}^{N+1}$ is computed by

$$c_i = \left. \frac{(z - p_i)b(z)}{za(z)} \right|_{z=p_i}, \quad 0 \leq i \leq N$$

where $b(z)$ is the input polynomial and $p_0 = 0$.

“If $c_i \neq 0$ for some $i > 0$, then the duration of the impulse response $h(k)$ is infinite, in which case S is an IIR system.”

IIR System: Example

Consider the system S governed by difference equation
 $y(k) - 0.2y(k-1) - 0.8y(k-2) = 2x(k) - 3x(k-1) + 4x(k-2)$

The characteristic polynomial of S is

$$a(z) = z^2 - 0.2z - 0.8 = (z-1)(z+0.8)$$

So, S has simple nonzero roots at $p = [1, -0.8]$.

Then, the impulse response is

$$h(k) = c_0\delta(k) + [c_1 + c_2(-0.5)^k]\mu(k)$$

The input polynomial is $b(z) = 2z^2 - 3z + 4$

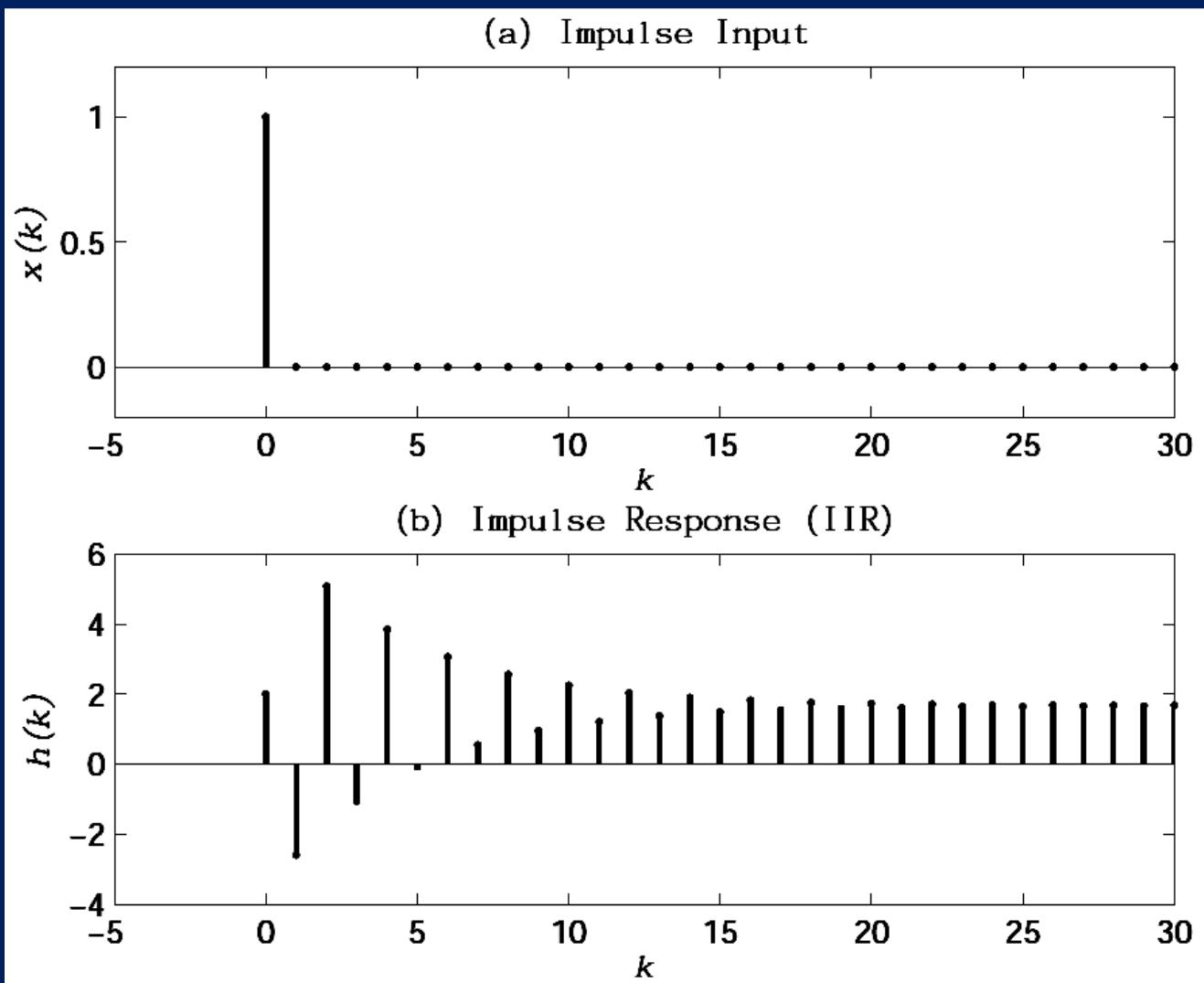
The coefficients are $c_0 = 4/[-1(0.8)] = -5$

$$c_1 = [2 - 3 + 4]/1.8 = 1.67$$

$$c_2 = [2(0.64) - 3(-0.8) + 4]/[(-0.8)(-1.8)] = 5.33$$

IIR System: Example

The impulse response of S has infinite duration. So, S is an IIR system.



Linear Convolution

Recall that $x(k) = \delta(k) \Rightarrow y(k) = h(k)$ (zero-state output)

For LTI system, $x(k) = \alpha\delta(k - i) \Rightarrow y(k) = \alpha h(k - i)$

“A causal signal $x(k)$ can be written as a weighted sum of unit impulses as”

$$x(k) = \sum_{i=0}^k x(i)\delta(k - i)$$

For LTI system, the zero-state output to the causal signal is

$$y(k) = \sum_{i=0}^k x(i)h(k - i)$$

The operation on the right-hand side is called **linear convolution** of signal $x(k)$ with signal $h(k)$:

$$x(k) * h(k) \stackrel{\text{def}}{=} \sum_{i=0}^k x(i)h(k - i)$$

Linear Convolution

Convolution is commutative:

$$\begin{aligned}x(k) * h(k) &= \sum_{i=0}^k x(i)h(k-i) \\&= \sum_{m=k}^0 x(k-m)h(m), \quad m = k-i \\&= \sum_{m=0}^k h(m)x(k-m) \\&= h(k) * x(k)\end{aligned}$$

For FIR system S , the impulse response is equal to the

input coefficient. So, $y(k) = \sum_{i=0}^k b(i)x(k-i)$

Properties of Linear Convolution

Name	Property
Commutative	$f \star g = g \star f$
Associative	$f \star (g \star h) = (f \star g) \star h$
Distributive	$f \star (g + h) = f \star g + f \star h$

Linear Convolution: Example

Consider the system

$$y(k) - 0.2y(k-1) - 0.8y(k-2) = 2x(k) - 3x(k-1) + 4x(k-2)$$

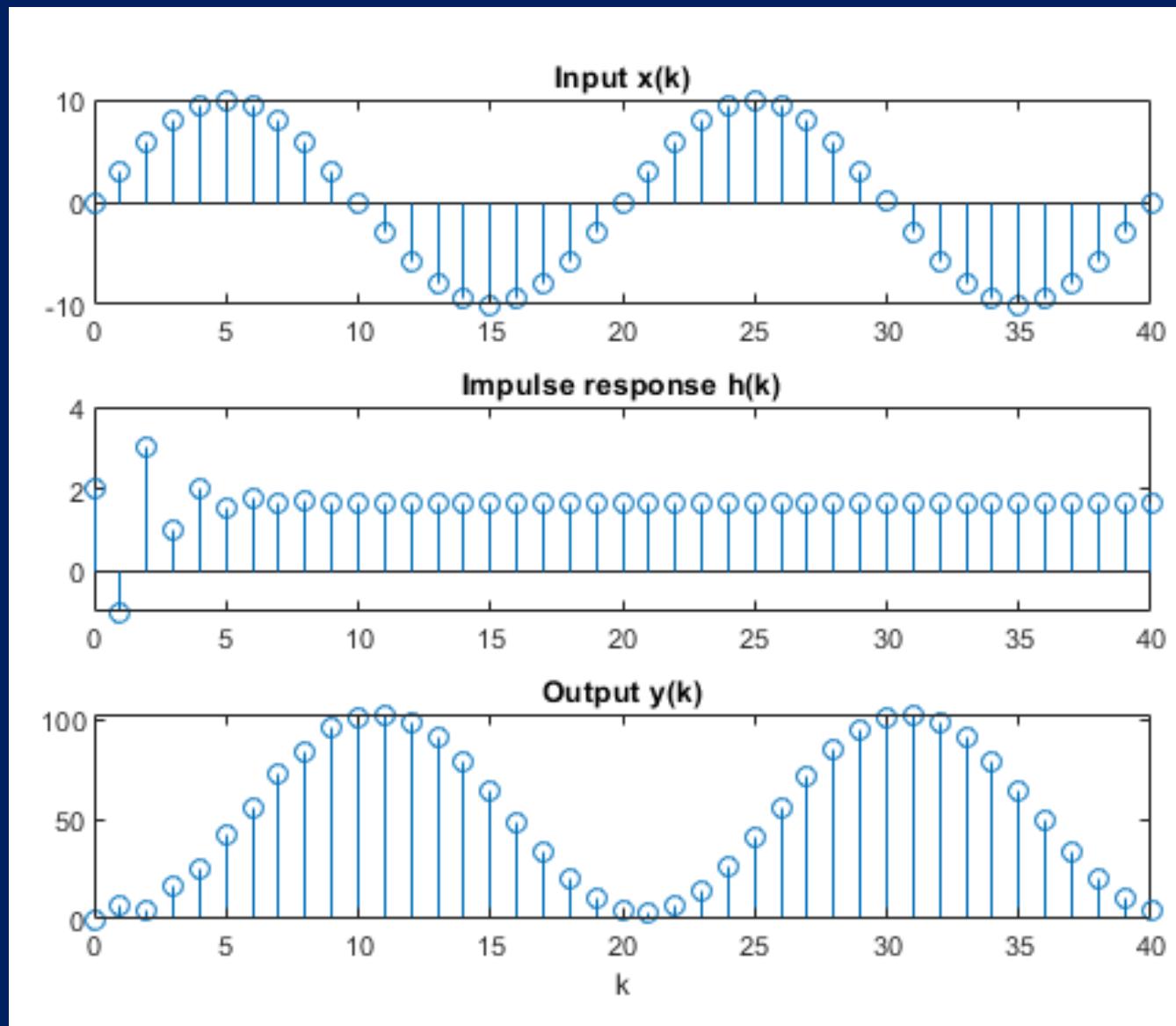
Suppose the input is $x(k) = 10 \sin(0.1\pi k) \mu(k)$

and the initial condition is zero. The output of the system
is then

$$\begin{aligned} y(k) &= \sum_{i=0}^k h(i)x(k-i) \\ &= \sum_{i=0}^k \left\{ -5\delta(i) + [1.67 - 5.33(-0.5)^i] \right\} \{10 \sin[0.1\pi(k-i)]\} \end{aligned}$$

MATLAB: `k = 0:40; x = 10*sin(0.1*pi*k);
y = filter([2, -3, 4], [1, -0.2, -0.8], x);`

Linear Convolution: Example



Linear Convolution

If $h(k)$ is nonzero for $0 \leq k < L$ and $x(k)$ is nonzero for $0 \leq k < M$, then the linear convolution can be expressed as

$$y(k) = \sum_{i=0}^{L-1} h(i)x(k-i), \quad 0 \leq k < L + M - 1$$

- “The upper limit has been changed from k to $L - 1$ because $h(i) = 0$ for $i \geq L$.”
- “Linear convolution of an L -point signal with an M -point signal is a signal of length $L + M - 1$.”

Linear Convolution

Linear convolution of L -point signal $h(k)$ and M -point signal $x(k)$ can be represented as matrix multiplication.

Example: $L = 2, M = 3$

$$\begin{bmatrix} y(0) \\ y(1) \\ y(2) \\ y(3) \end{bmatrix} = \underbrace{\begin{bmatrix} x(0) & 0 \\ x(1) & x(0) \\ x(2) & x(1) \\ 0 & x(2) \end{bmatrix}}_{\text{linear convolution matrix}} \begin{bmatrix} h(0) \\ h(1) \end{bmatrix}$$

linear convolution matrix

Circular Convolution

- Circular convolution is an operator whose result has the same length as the two operands.
- “The periodic extension of an N -point signal $x(k)$ is a signal $x_p(k)$ defined for all integers k as follows.”

$$x_p(k) \stackrel{\text{def}}{=} x(k \text{ modulo } N)$$

where the modulo operator returns the remainder of the division between two numbers. Example: 5 modulo 2 = 1

MATLAB: `mod(5, 2)`

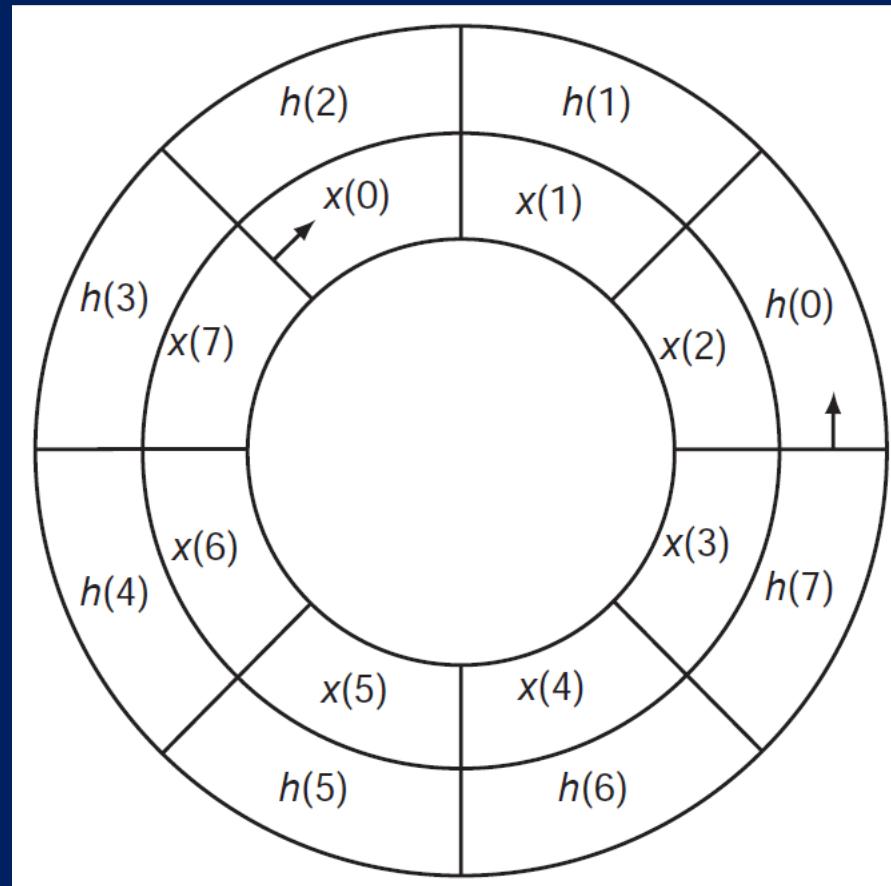
- “ $x_p(k)$ extends $x(k)$ periodically in both positive and negative directions.”
- Example: $x_p(N) = x(0)$, $x_p(-1) = x(N-1)$

Circular Convolution

Circular convolution of N -point signals $h(k)$ and $x(k)$ is defined as

$$h(k) \circ x(k) \stackrel{\text{def}}{=} \sum_{i=0}^{N-1} h(i)x_p(k-i), \quad 0 \leq k < N$$

Example: $k = 2, N = 8$



Circular Convolution

Circular convolution of N -point signals $h(k)$ and $x(k)$ can be represented as matrix multiplication.

Example: $N = 4$

$$\begin{bmatrix} y(0) \\ y(1) \\ y(2) \\ y(3) \end{bmatrix} = \underbrace{\begin{bmatrix} x(0) & x(3) & x(2) & x(1) \\ x(1) & x(0) & x(3) & x(2) \\ x(2) & x(1) & x(0) & x(3) \\ x(3) & x(2) & x(1) & x(0) \end{bmatrix}}_{\text{Circular convolution matrix}} \begin{bmatrix} h(0) \\ h(1) \\ h(2) \\ h(3) \end{bmatrix}$$

Zero Padding

- Circular convolution can be implemented using the fast Fourier transform (FFT). However, circular convolution produce a different response than linear convolution (zero-state response of a linear discrete-time system).
- Zero padding can be used to modify the L -point signal $h(k)$ and M -point signal $x(k)$ so that circular convolution provides the same result as linear convolution.

$$h_z = [h(0), h(1), \dots, h(L-1), \overbrace{0, \dots, 0}^{M-1}]^T$$

$$x_z = [x(0), x(1), \dots, x(M-1), \underbrace{0, \dots, 0}_{L-1}]^T$$

Zero Padding

“Thus, h_z and x_z are zero-padded vectors of length $N = L + M - 1$. ”

$$\begin{aligned} h_z(k) \circ x_z(k) &= \sum_{i=0}^{N-1} h_z(i) x_{zp}(k-i), & 0 \leq k < N \\ &= \sum_{i=0}^{L-1} h_z(i) x_{zp}(k-i) & x_{zp}(k) \text{ is periodic extension of } x_z(k). \\ &= \sum_{i=0}^{L-1} h_z(i) x_z(k-i) & h_z(k) \text{ has only } L-1 \text{ nonzero elements} \\ &= h_z * x_z = h * x & x_z(k) \text{ has only } L-1 \text{ zeros padded to} \\ & & \text{the end of it. So, } x_{zp}(k-i) = x_z(k-i) \\ & & \text{This can be easily verified.} \end{aligned}$$

$$\text{So, } h_z(k) \circ x_z(k) = h(k) * x(k), \quad 0 \leq k < N$$

Exercise

Let $x = \begin{bmatrix} 1 \\ 2 \end{bmatrix}$, $h = \begin{bmatrix} 1 \\ 2 \\ 3 \end{bmatrix}$. Verify that $h * x = h_z \circ x_z$.

Deconvolution

- When the impulse response $h(k)$ and the output $y(k)$ of the system are known, the input $x(k)$ is to be obtained.
- This process is referred to as deconvolution.
- Suppose $h(k)$ and $x(k)$ are both causal and $h(0) \neq 0$.
- Evaluating $y(k) = \sum_{i=0}^k x(i)h(k-i)$ at $k = 0$ yields
$$y(0) = x(0)h(0) \Rightarrow x(0) = y(0)/h(0)$$
- Once $x(0)$ is known, the remaining samples of $x(k)$ can be obtained recursively.
- At $k = 1$, $y(1) = x(0)h(1) + x(1)h(0)$. Solving for $x(1)$ yields
$$x(1) = \frac{y(1) - x(0)h(1)}{h(0)}$$

Deconvolution

- Repeating the process for $2 \leq k < N$, we obtain

$$x(k) = \frac{1}{h(0)} \left[y(k) - \sum_{i=0}^{k-1} x(i)h(k-i) \right], \quad k \geq 1$$

- “Deconvolution also includes finding the impulse response $h(k)$, given the input $x(k)$ and the output $y(k)$.” This is a special case of **system identification**.

Exercise: Given the impulse response and the output of a system as $h = [2, -1, 6]^T$, $y = [10, 1, 19, 22, -24]^T$. Find the input $x(k)$.

Polynomial Arithmetic

“Suppose $A(z)$ and $B(z)$ are polynomials of degree L and M , respectively.”

$$A(z) = a_0 z^L + a_1 z^{L-1} + \cdots + a_L$$

$$B(z) = b_0 z^M + b_1 z^{M-1} + \cdots + b_M$$

Let $C(z) = A(z)B(z)$ be the product polynomial of degree $N = L + M$, that is $C(z) = c_0 z^{L+M} + c_1 z^{L+M-1} + \cdots + c_{L+M}$

The coefficients of $C(z)$ can be obtained using linear convolution between the coefficients of $A(z)$ and $B(z)$ as

$$c(k) = a(k) * b(k), \quad 0 \leq k < L + M + 2$$

where $a(k)$, $b(k)$, $c(k)$ are the coefficients of polynomials $A(z)$, $B(z)$, $C(z)$.

Polynomial Arithmetic: Example

Consider the polynomials

$$A(z) = 2z^2 - z + 6, \quad B(z) = 5z^2 + 3z - 4$$

whose coefficient vectors are $a = [2, -1, 6]^T, b = [5, 3, -4]^T$

Then, $c(k) = a(k) * b(k) = [10, 1, 19, 22, -24]^T$

As a result, $C(z) = A(z)B(z) = 10z^4 + z^3 + 19z^2 + 22z - 24$

MATLAB:

```
a = [2, -1, 6];  
b = [5, 3, -4];  
c = conv(a, b);
```

Linear Cross-Correlation

Linear cross-correlation of L -point signal $y(k)$ with M -point signal $x(k)$ where $M \leq L$ is denoted as $r_{yx}(k)$ and defined as

$$r_{yx}(k) = y(k) \otimes x(k) \stackrel{\text{def}}{=} \frac{1}{L} \sum_{i=0}^{L-1} y(i)x(i-k), \quad 0 \leq k < L$$

- If $x(k)$ is causal, the lower limit can be set to $i = k$.
- “Linear cross-correlation is sometimes defined without the scaling factor $1/L$.”
- “The scaling factor is used here so that it is consistent with the statistical definition of cross-correlation between two random signals.”

Linear Cross-Correlation

Linear cross-correlation of L -point signal $y(k)$ with M -point signal $x(k)$ can be represented as matrix multiplication.

Example: $L = 3, M = 2$

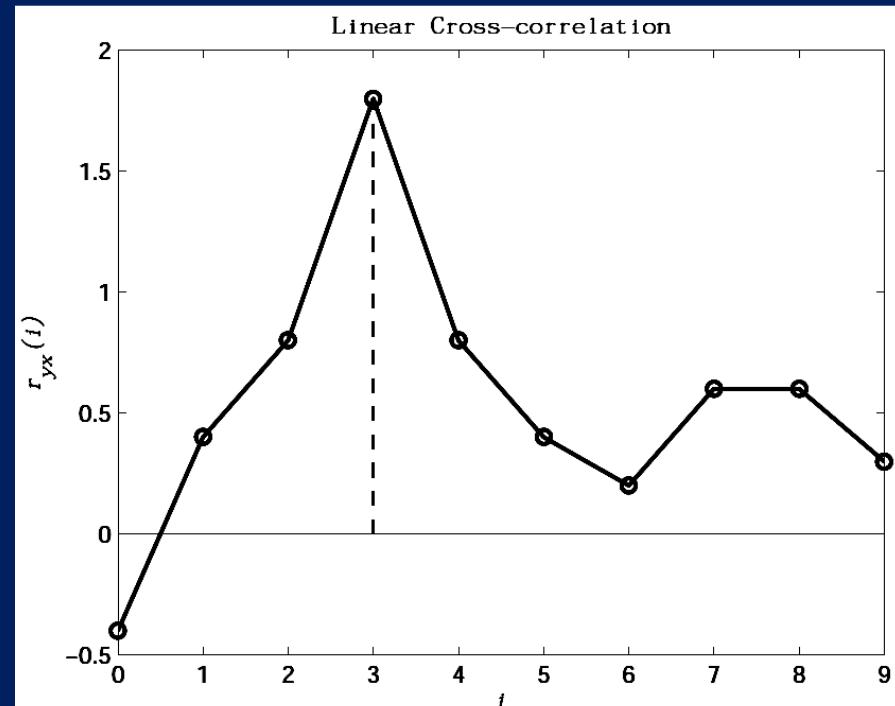
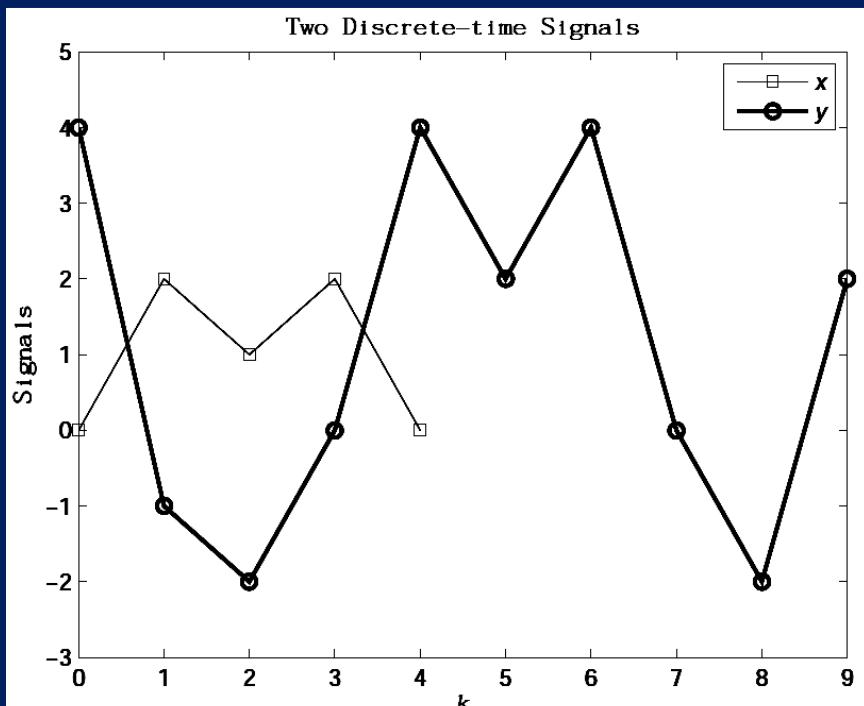
$$\begin{bmatrix} r_{yx}(0) \\ r_{yx}(1) \\ r_{yx}(2) \end{bmatrix} = \underbrace{\frac{1}{3} \begin{bmatrix} x(0) & x(1) & 0 \\ 0 & x(0) & x(1) \\ 0 & 0 & x(0) \end{bmatrix}}_{\text{linear cross-correlation matrix}} \begin{bmatrix} y(0) \\ y(1) \\ y(2) \end{bmatrix}$$

Thus, linear cross-correlation of $y(k)$ with $x(k)$ can be expressed as $r_{yx} = D(x)y$ where $D(x)$ is correlation matrix.

Exercise

“Linear cross-correlation can be used to measure the degree to which the shape of one signal is similar to the shape of the other signal.”

Given $x = [0, 2, 1, 2, 0]^T$, $y = [4, -1, -2, 0, 4, 2, 4, 0, -2, 2]^T$,
Compute the linear cross-correlation of $y(k)$ with $x(k)$.



Normalized Linear Cross-Correlation

Proakis and Manolakis (1992) showed that the square of cross-correlation is bounded

$$r_{yx}^2(k) \leq \left(\frac{M}{L} \right) r_{xx}(0) r_{yy}(0), \quad 0 \leq k < L$$

The normalized linear cross-correlation is then defined as

$$\rho_{yx}(k) \stackrel{\text{def}}{=} \frac{r_{yx}(k)}{\sqrt{(M/L)r_{xx}(0)r_{yy}(0)}}, \quad 0 \leq k < L$$

As a result,

$$-1 \leq \rho_{yx}(k) \leq 1, \quad 0 \leq k < L$$

When a correlation peak approaches ± 1 , there is a strong correlation between $y(k)$ and $x(k)$.

Circular Cross-Correlation

“Let $y(k)$ and $x(k)$ be N -point signals, and let $x_p(k)$ be the periodic extension of $x(k)$. The circular cross-correlation of $y(k)$ with $x(k)$ is denoted as $c_{yx}(k)$ and defined as”

$$c_{yx}(k) \stackrel{\text{def}}{=} \frac{1}{N} \sum_{i=0}^{N-1} y(i)x_p(i-k), \quad 0 \leq k < N$$

Normalized circular cross-correlation is then defined as

$$\sigma_{yx}(k) \stackrel{\text{def}}{=} \frac{c_{yx}(k)}{\sqrt{c_{xx}(0)c_{yy}(0)}}$$

Properties of Cross-Correlation

Symmetry: $c_{xy}(k) = c_{yx}(-k), \quad 0 \leq k < N$

Relationship with convolution:

$$c_{yx}(k) = \frac{y(k) \circ x(-k)}{N}, \quad 0 \leq k < N$$

Relationship between linear and circular correlations:
 $y(k)$ = L -point signal, $x(k)$ = M -point signal, $M \leq L$.

$$r_{yx}(k) = \left(\frac{N}{L} \right) c_{y_z x_z}(k), \quad 0 \leq k < L$$

where $y_z(k)$ and $x_z(k)$ are zero-padded versions of $y(k)$ and $x(k)$, respectively, such that both $y_z(k)$ and $x_z(k)$ are of length $L + M + p$ with $p \geq -1$.

Lab: Echo Detection

- Let $x(k)$ be transmitted signal and $y(k)$ be received signal.
- Suppose the sampling frequency is $f_s = 1$ MHz and the number of transmitted samples is $M = 512$.
- Let $x(k)$ be a multi-frequency chirp, a sinusoidal signal whose frequency varies with time:

$$f(k) = kf_s / [2(M - 1)]$$

$$y(k) = \sin[2\pi f(k)kT], \quad T = 1/f_s$$

- “The received signal $y(k)$ includes a scaled and delayed version of the transmitted signal plus measurement noise.”

Lab: Echo Detection

- Suppose received signal consists of $L = 2048$ samples.
- “If $x_z(k)$ denotes the transmitted signal, zero-extended to L points, then the received signal can be expressed as”

$$y(k) = ax_z(k - d) + \eta(k), \quad 0 \leq k < L$$

- The first term is the echo of the transmitted signal with attenuation factor $a \ll 1$ and delayed by d samples.
- The second term is atmospheric noise. Suppose the noise is uniformly distributed over the interval $[-\sigma, \sigma]$.
- If T is the sampling interval, then the time of flight in seconds is $\tau = dT$. Distance to the target is $r = cdT/2$ where c is the speed of light and the factor 2 arises because the time of flight is a two-way travel time.

Lab: Echo Detection

- Generate the input signal $x(k)$, atmospheric noise $\eta(k)$, and received signal $y(k)$ with your own choice of attenuation factor a , delay d , and noise bound σ .
- Perform linear cross-correlation and normalized linear cross-correlation of $y(k)$ with $x(k)$ to determine the delay d . Also plot the correlation results.

References

- Schilling, R. J. and S. L. Harris, 2012, Fundamentals of Digital Signal Processing using MATLAB, Second Edition, Cengage Learning.
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