

Discrete-time signals and systems, II

- Discrete-time systems
- Representation of discrete-time systems
- Classification of discrete-time systems
- Discrete Linear Time-Invariant (LTI) systems
- Convolution
- Convolution, summary
- Convolution properties
- Examples of LTI systems
- Impulse response and stability
- (Additional materials)



Discrete-time systems

• A *discrete-time* (DT) *system* is a device or algorithm that operates on a discrete-time signal, x(n), called the *input* or *excitation*, according to some well-defined rule, to produce another discrete-time signal, y(n), called the *output* or *response* of the system

$$x(n)$$
 DT system (T) $y(n) = T [x(n)]$

- The input-output description of a discrete-time system consists of a mathematical expression or a rule, which explicitly defines the relation between the input and output signals (T – inputoutput relationship)
 - \rightarrow Difference equation (DE)
- Examples of discrete-time systems:
 - Delaying the input y(n) = x(n-5)
 - Moving average filter y(n) = 1/3 [x(n+1) + x(n) + x(n-1)]
 - Squaring the input y(n) = x(n) x(n)
 - Minimum $y(n) = \min \{ x(n+1), x(n), x(n-1) \}$

Representation of discrete-time systems

- Difference equation (DE) of a system, y(n) = T[x(n), x(n 1), x(n 2), ...]
 - Example

$$y(n) = \frac{1}{4} [x(n) + x(n-1) + x(n-2) + x(n-3)] = \frac{1}{4} \sum_{k=n-3}^{n} x(k)$$

• Block diagram realization of the system



- What is the output y(n), if the input is rectangular signal, $x(n) = \{1, 1\}$?

Representation of discrete-time systems

• Example
$$y(n) = \frac{1}{4} [x(n) + x(n-1) + x(n-2) + x(n-3)] = \frac{1}{4} \sum_{k=n-3}^{n} x(k)$$



 $y(n) = \{ 1/4, 1/2, 1/2, 1/2, 1/4 \}$

- Linear versus nonlinear systems
 - Linear systems obey superposition, i.e., additivity and homogeneity (scaling)



- Linear versus nonlinear systems
 - A system T



is linear if and only if

 $T[a_1x_1(n) + a_2x_2(n)] = a_1T[x_1(n)] + a_2T[x_2(n)]$

for all $x_1(n)$, $x_2(n)$, and all constants a1 and a2 (additivity and scaling)

- How to determine if the system is linear?

- 1. $y_1(n) = T [x_1(n)]$
- 2. $y_2(n) = T [x_2(n)]$
- 3. $T[a_1x_1(n) + a_2x_2(n)] = a_1y_1(n) + a_2y_2(n)$

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Classification of discrete-time systems $y(n) = \sum x(l)$ • Linearity, example, accumulator \rightarrow 1. $y_1(n) = \sum_{l=-\infty} x_1(l)$ 2. $y_2(n) = \sum_{l=-\infty}^n x_2(l)$ $x(n) = a_1 x_1(n) + a_2 x_2(n) \rightarrow$ Linear combination of inputs 3. \longrightarrow $y(n) = \sum (a_1 x_1(l) + a_2 x_2(l))$ T[x(n)] $y(n) = \sum_{l=1}^{n} (a_1 x_1(l)) + \sum_{l=1}^{n} (a_2 x_2(l))$ $y(n) = a_1 \sum_{l=-\infty}^{n} x_1(l) + a_2 \sum_{l=-\infty}^{n} x_2(l) \rightarrow y_1(n) = \sum_{l=-\infty}^{n} x_1(l)$ $\underbrace{\longrightarrow} y_2(n) = \sum x_2(l)$ Linear

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- Time-invariant systems versus time-variant systems
 - Time shift of input causes the same shift at output



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• Time-invariant systems versus time-variant systems

- A system T is time-invariant (or shift-invariant) if and only if

$$y(n) = T [x(n)]$$

implies that

$$y(n - n_0) = T[x(n - n_0)]$$

for every input signal x(n) and every time shift n_0)

- How to determine if the system is time-invariant?

- 1. y(n) = T [x(n)]
- 2. $y_1(n, n_0) = T[x(n n_0)]$ (delay the input, x(n))
- 3. $y(n n_0) = T [x(n n_0)]$ (delay the output, *n*)

4.
$$y_1(n, n_0) = y(n - n_0)$$

• Time-invariant, example, \rightarrow (1.) $y(n) = (x(n))^2$

Delay the input 2. $y_1(n, n_0) = (x(n-n_0))^2$ Delay the output 3. $y(n-n_0) = (x(n-n_0))^2$

4.
$$y_1(n, n_0) = y(n-n_0)$$

• Time-variant, example, \rightarrow (1.) $y(n) = n \cdot x(n)$

4.

 Delay the input
 2.
 $y_1(n, n_0) = n \cdot x(n-n_0)$

 Delay the output
 3.
 $y(n-n_0) = (n-n_0) \cdot x(n-n_0)$

$$y_1(n, n_0) \neq y(n-n_0)$$

- Example of exam task
 - Is the following system linear? Is it time-invariant?

$$y(n) = x^{2}(n) - x(n-1) \cdot x(n+1)$$

• Causal versus non-causal systems

- A system is called causal if the output of the system at any time *n* depends only on present and past inputs, but does not depend on future inputs

y(n) = T[x(n), x(n-1), x(n-2), ...]

- A causal system is physically realizable system (not for off-line processing)
- (A signal is called causal if and only if x(n) = 0 for n < 0)
- Backward difference, $y(n) = x(n) x(n-1) \rightarrow causal$
- Forward difference, $y(n) = x(n+1) x(n) \rightarrow non-causal$, looks forward in time
- Causality \rightarrow consequence happens after the cause

- Check for causality
- Moving average

$$y(n) = \frac{1}{M} \sum_{k=0}^{M-1} x(n-k)$$

 \rightarrow causal, y(n) depends on x(n - k), $k \ge 0$,

• Centered moving average (counter example)

$$y(n) = \frac{1}{M} \sum_{k=-(M-1)/2}^{(M-1)/2} x(n-k)$$

 \rightarrow non-causal, looks forward in time, can be made causal by delaying

• Stable versus unstable systems

- A system is said to be **Bounded Input Bounded Output (BIBO) stable** if and only if every bounded input produces a bounded output
- There exist some finite numbers Bx and By such that

$$\exists B_x, B_y: |x(n)| \leq B_x < \infty, |y(n)| \leq B_y < \infty$$

Discrete Linear Time-Invariant (LTI) systems

• An important subset of discrete systems – discrete Linear Time-Invariant Systems (LTI)

- Properties

- * Linearity (allows analysis using elementary signals)
- * Time-invariability (operating does not change over time)
- Consequences
 - * Mathematical tractability
 - * Simple realization
 - $^* \rightarrow \text{Convolution}$



• Why convolution?

So far, we were able to determine

- the output, y(n), an LTI system to the input signal, x(n),
- if knowing
 - * the difference equation
 - * or, block diagram of the LTI system

of
$$y(n) = \frac{1}{4} [x(n) + x(n-1) + x(n-2) + x(n-3)] = \frac{1}{4} \sum_{k=n-3}^{n} x(k)$$



 $x(n) \rightarrow T[] \rightarrow y(n)$

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• Why convolution?

Using convolution, we can determine

- the output, y(n), of an LTI system to the input signal, x(n),
- if knowing

 the response of
 the LTI system
 (denoted h(n)) to one
 particular input signal,
 the unit sample signal, δ(n)



If $x(n) = \delta(n)$ then y(n) = h(n)

 $\delta(n)$ - the unit sample h(n) - impulse response

 $x(n), h(n) \rightarrow convolution \rightarrow y(n)$

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- Impulse response
- Difference equation, y(n) = ?

$$x(n) = \begin{bmatrix} 1 \\ -3 \\ -2 \\ -1 \end{bmatrix} = \begin{bmatrix} 1 \\ 1 \\ 2 \\ 3 \\ 4 \\ 5 \\ 6 \\ 7 \end{bmatrix}$$

- Output values, y(n) = ?



$$x(n) = \delta(n)$$
 impulse - unit sample
 \downarrow
 $y(n) = h(n)$ impulse response = ?

$$y(n) = h(n) = ?$$

• Impulse response

 $y(n) = A_{1.x}(n) + A_{2.x}(n-1) + A_{3.x}(n-2)$





 $x(n) = \delta(n) \text{ impulse - unit sample}$ y(n) = h(n) impulse response $h(n) = A_1 \cdot \delta(n) + A_2 \cdot \delta(n-1) + A_3 \cdot \delta(n-2)$ $y(n) = h(n) = \{A_1, A_2, A_3\}$ n = 0

 \rightarrow LTI system is completely specified by h(n)

• Decomposition of discrete signal x(n)

- Discrete signal x(n) can be expressed by a sum of unit samples

$$x(n) = \sum_{k=-\infty}^{\infty} x(k) \,\delta(n-k)$$

where

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

- Example, <u>causal signal</u>: x(n) = 0, n < 0

$$x(n) = x(0)\delta(n) + x(1)\delta(n-1) + x(2)\delta(n-2) + \dots$$



• Response of an LTI system

- Suppose that h(n) is impulse response to unit sample $\delta(n)$, $\delta(n) \rightarrow T[] \rightarrow h(n)$
- Since the system is

time-invariant
$$\longrightarrow T[\delta(n)] = h(n) \Rightarrow T[\delta(n-k)] = h(n-k)$$

and linear $\longrightarrow T[a. \delta(n-k)+b. \delta(n-l)] = a.h(n-k)+b.h(n-l)$

- The output is as follows:

$$y(n) = T[x(n)]$$

$$= T \left[\sum_{k=-\infty}^{\infty} x(k) \delta(n-k)\right]$$

$$= \sum_{k=-\infty}^{\infty} x(k) T[\delta(n-k)] \longrightarrow \text{Due to linearity}$$

$$= \sum_{k=-\infty}^{\infty} x(k) h(n-k) \longrightarrow \text{Due to definition of } h(n) \text{ and time-invariance}$$

$$y(n) = \sum_{k=-\infty}^{\infty} x(k) h(n-k) = x(n) * h(n) \longrightarrow \text{Convolution sum}$$
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[Oppenheim, Schafer]





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Convolution

- Interpretation / implementation of convolution
 - Convolving a signal with an impulse response of an LTI system is equivalent to passing the signal through the system (summation is symmetric in x and h)

$$y(n) = \sum_{k=-\infty}^{\infty} x(k) h(n-k) = \sum_{k=-\infty}^{\infty} h(k) x(n-k)$$
1. The system is moving
over the signal
("Off-line" analysis)
Flipped impulse response, h(n)
$$("Off-line" analysis)$$
Input signal is passing
the system
("Real-time" analysis)
Input signal, x(n)
Flipped x(n)
Input signal, x(n)
Flipped x(n)
Impulse response, h(n)
Impulse response, h(

Demo: https://engineering.purdue.edu/VISE/ee438/demos/



- Example
- A block diagram of an LTI system, y(n) = T[x(n), y(n 1)]



- Difference equation: y(n) = 0.25 y(n 1) + 0.5 x(n) + 0.5 x(n 1)
- What is the length of impulse response?
- Convolution ?

→ Convolution is applicable only for those LTI systems for which y(n) = T[x(n), x(n-1), ..., x(n-M)]
 i.e., impulse response, h(n), is of finite duration



- Example of exam task
 - An LTI system is defined by the following impulse response

 $h(n) = 2 \cos(n \pi / 2) \cdot [u(n) - u(n - 3)]$

Determine the response of the system using convolution for the following input signal

x(n) = n . [u(n) - u(n - 4)]

Hint: Consider the difference of two unit step signals, u(.)



Convolution, summary

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



- The impulse response h(n) can determine the characteristics of an LTI system completely
- For any input x(n), the output y(n) is the convolution sum of x(n) and h(n)
- Convolution can be used to implement only those LTI systems for which the impulse response *h*(*n*) is of finite duration

$$h(n) = \begin{cases} \neq 0, & 0 \leq n < N \\ 0, & \text{elsewhere} \end{cases}$$

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Convolution properties

- Convolution properties define connections of LTI systems
 - Convolution is a commutative operation

x(n) * h(n) = h(n) * x(n)



- Convolution is an associative operation \rightarrow Cascade connection

 $[x(n) * h_1(n)] * h_2(n) = x(n) * [h_1(n) * h_2(n)]$



again commutativity and again associativity \rightarrow Order of systems exchanged = $x(n) * [h_2(n) * h_1(n)] = [x(n) * h_2(n)] * h_1(n)$

$$x(n) \longrightarrow h_2(n) * h_1(n) \longrightarrow y(n) \iff x(n) \longrightarrow h_2(n) \longrightarrow h_1(n) \longrightarrow y(n)$$
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Convolution properties

- Convolution properties define connections of LTI systems
 - Convolution is distributive over addition \rightarrow Parallel connection
 - $x(n) * h_1(n) + x(n) * h_2(n) = x(n) * [h_1(n) + h_2(n)]$





Convolution properties

- Example of exam task
 - $y_1 = x_1(n) x_1(n-1)$, $y_2 = x_2(n) + x_2(n-1)$; Derive $h_1(n)$, $h_2(n)$, and $h_c(n)$ for cascade and $h_p(n)$ for parallel connection. Check commutativity for $h_1(n)$ and $h_2(n)$

• Example of exam task

- Check the properties of output signal for cascade and parallel connection in the following case: Fs = 6000 smp/sec, h1 passes frequencies from 0 – 250 Hz, h2 passes from 1700 – 2400 Hz



- Example of finite impulse response LTI system, simple lowpass filter
- $y_{L}(n) = 1/2 x(n) + 1/2 x(n-1)$
 - \rightarrow * Moving average (first order),
 - * Equalizes values of neighboring samples,
 - * Lowpass filtering effect,
 - * Keeps slowly varying trend of values of distant samples
 - * Rejects rapid changes of values of neighboring samples



\rightarrow Smoothing effect

$$h_{L}(n) = 1/2 \,\delta(n) + 1/2 \,\delta(n-1)$$

 $h_{L}(n) = \{1/2, 1/2\}$



- Example of finite impulse response LTI system, simple lowpass filter
- Repetitive use of simple (first-order) moving average lowpass filter, hL(n), will increase the smoothing effect

Since convolution is an associative operation

[x(n) * hL(n)] * hL(n) = x(n) * [hL(n) * hL(n)]

$$x(n) \longrightarrow h_{L}(n) \longrightarrow h_{L}(n) \longrightarrow y(n) \iff x(n) \longrightarrow h_{L}(n) * h_{L}(n) \longrightarrow y(n)$$

follows that the impulse response of the (second-order) moving average filter, $h_{L2}(n)$, which will increase the lowpass filtering effect, could be obtained by convolving $h_{H}(n)$ by itself

 $h_{L2}(n) = h_{L}(n) * h_{L}(n) = \{1/2, 1/2\} * \{1/2, 1/2\} = \{1/4, 1/2, 1/4\}$

Verify the lowpass filtering effect of hL2(n) at <u>http://www.fourier-series.com/</u>

 \rightarrow hL2(n) - principal impulse response for smoothing spatial filters (*image smoothing*)



- Example of finite impulse response LTI system, simple highpass filter
- $y_H(n) = 1/2 x(n) 1/2 x(n-1)$
 - \rightarrow * Differentiator (first order),
 - * Preserves differences of values of neighboring samples,
 - * Highpass filtering effect,
 - * Keeps rapid changes of values of neighboring samples
 - * Rejects slowly varying trend of values of distant samples



 \rightarrow Sharpening effect

hh(n) = $1/2 \delta(n) - 1/2 \delta(n - 1)$ hh(n) = $\{1/2, -1/2\}$



- Example of finite impulse response LTI system, simple highpass filter
- Repetitive use of simple first-order differentiator highpass filter, hH(n), will increase the sharpening effect

Since convolution is an associative operation

[x(n) * hH(n)] * hH(n) = x(n) * [hH(n) * hH(n)]

$$x(n) \longrightarrow h_H(n) \longrightarrow y(n) \longrightarrow x(n) \longrightarrow h_H(n) * h_H(n) \longrightarrow y(n)$$

follows that the impulse response of the second-order differentiator filter, $h_{H2}(n)$, which will increase the highpass filtering effect, could be obtained by convolving $h_{H}(n)$ by itself

 $h_{H2}(n) = h_{H}(n) * h_{H}(n) = \{1/2, -1/2\} * \{1/2, -1/2\} = \{1/4, -1/2, 1/4\}$

Verify the highpass filtering effect of hH2(n) at <u>http://www.fourier-series.com/</u>

 \rightarrow hH2(n) - principal impulse response for sharpening spatial filters (*image sharpening*)



• Further study, laboratory

1) Using the impulse response, $h_{L}(n) = \{1/2, 1/2\}$ of simple moving average lowpass filter, $y_{L}(n) = 1/2 x(n) + 1/2 x(n - 1)$, derive the impulse response of the fourth order, i.e., $h_{L}(n) = h_{L}(n) * h_{L}(n) * h_{L}(n) * h_{L}(n)$

Verify lowpass filtering effect of the filter $h_{L4}(n)$ at

http://www.fourier-series.com/

2) Using the impulse response, $h_{H}(n) = \{1/2, -1/2\}$ of simple differentiator highpass filter, $y_{H}(n) = 1/2 x(n) - 1/2 x(n - 1)$, derive the impulse response of the fourth order, i.e., $h_{H4}(n) = h_{H}(n) * h_{H}(n) * h_{H}(n)$

Verify highpass filtering effect of the filter $h_{H4}(n)$ at

http://www.fourier-series.com/

Impulse response and stability

• An LTI system is said to be **Bounded Input Bounded Output (BIBO) stable** if and only if every bounded input produces a bounded output

$$B_{x}, B_{y}: |x(n)| \leq B_{x} < \infty, |y(n)| \leq B_{y} < \infty$$

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

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

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

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

• An LTI system is stable, $B_{y} < \infty$, if and only if

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- $\sum_{k=-\infty}^{\infty} |h(k)| < \infty$
- The system can give meaningful output only if it is stable



(Additional materials)

- Implementation of convolution
- Convolution, matrix interpretation
- Convolution, example
- Convolution, example
- Convolution in MATLAB
- Convolution and causality



Implementation of convolution

• Convolution sum for LTI systems

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

• Summation is symmetric in x and h

i.e.
$$l = n-k \Rightarrow$$

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

Demo: <u>https://engineering.purdue.edu/VISE/ee438/demos/</u>

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Convolution, matrix interpretation

• Convolution interpretation 2, matrix interpretation

$$y(n) = \sum_{k=-\infty}^{\infty} h(k) x(n-k) = \sum_{k=-\infty}^{\infty} x(n-k)h(k)$$
$$\begin{bmatrix} y(0) \\ y(1) \\ y(2) \\ \dots \end{bmatrix} = \begin{bmatrix} x(0) x(-1) x(-2) \\ x(1) x(0) x(-1) \\ x(2) x(1) x(0) \\ \dots \dots \end{bmatrix} \begin{bmatrix} h(0) \\ h(1) \\ h(2) \end{bmatrix}$$

Diagonals in X matrix are equal



Convolution, example

• Convolution example, inverting attenuator and discrete derivative

a. Inverting Attenuator



Demo: <u>http://www.fourier-series.com/</u> Course: 63744

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Convolution, example

• Example

Derive (analytically) convolution for the following example

- Rectangular pulse of duration N

$$x[n] = u[n] - u[n - N] = \begin{cases} 1 & \text{if } 0 \le n \le N - 1\\ 0 & \text{otherwise} \end{cases}$$

is passed to the first-order low-pass filter of which impulse response is

 $h[n] = a^n u[n]$

Solution: For n < 0

$$n[n] = 0$$

For
$$n \le 0 \le N - 1$$

$$y[n] = \sum_{m=0}^{n} a^m = \frac{1 - a^{n+1}}{1 - a}$$

For
$$n \ge N$$

$$y[n] = \sum_{m=n-N+1}^{n} a^m = a^n - N + 1 \frac{1 - a^N}{1 - a} = \frac{1 - a^- N}{1 - a^- 1} a^n$$

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Convolution, example





Convolution example: (A) Unit-sample response of the first-order low-pass filter, $h[n] = a^n u[n]$, with a = 0.9. (B) Input signal, x[n] = u[n] - u[n - N], with N = 10. (C) Output signal.



Convolution in MATLAB

• Convolution in MATLAB

- The function **conv()** implements the convolution sum of two finite-length sequences



Convolution and causality

An LTI system is causal if and only if h(n) = 0 for n < 0
Convolution

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

- Causal LTI system (h(n) causal) and causal input signal x(n)

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