

Linear Time Invariant and Causal (LTIC) Systems

Digital Signal Processing

Notes

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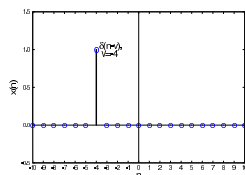
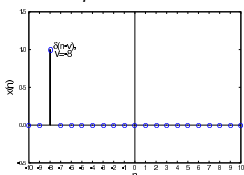
Notes

Unit Impulse Function

- Unit impulse function is a fundamental function in Digital Signal Processing (DSP)
- Symbol of Unit impulse function is the *Greek delta*: δ
- $\delta(n) = 1$ if $n = 0$, so that,

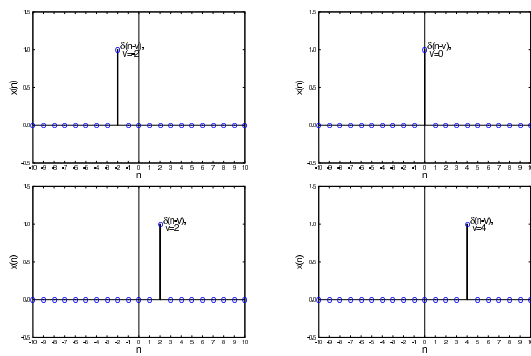
$$\delta(n - v) = \begin{cases} 1 & \text{if } (n - v) = 0, \\ 0 & \text{otherwise.} \end{cases}$$

■ Examples



Notes

Unit Impulse Function *cont'd.*



Notes

Scaling Unit Impulse Function

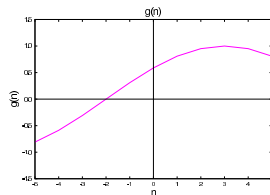
- Can scale unit impulse with any value, *i.e.*

$$g \times \delta(n-v) = \begin{cases} g & \text{if } n-v=0, \\ 0 & \text{otherwise.} \end{cases}$$

- So if g is a function, such as $g(n)$ then

$$g(n)\delta(n-v) = \begin{cases} g(n) & \text{if } n-v=0, \\ 0 & \text{otherwise.} \end{cases}$$

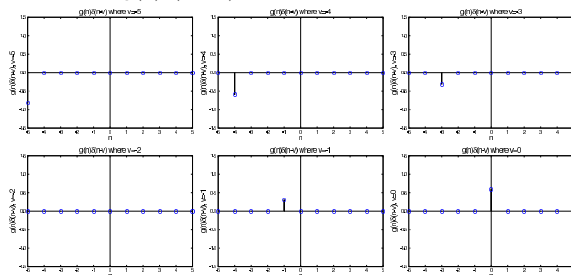
- This is useful for something called *sifting*
- Given a signal $g(n)$:



Notes

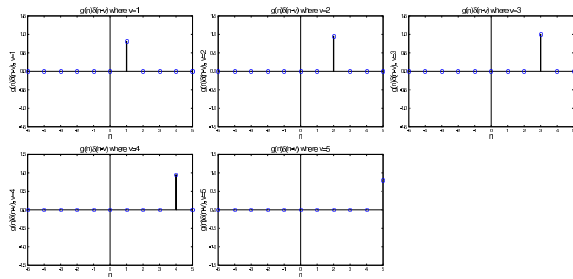
Sifting

- Calculate $g(n)\delta(n-v)$ for all values of v , *i.e.*



Notes

Sifting *cont'd.*

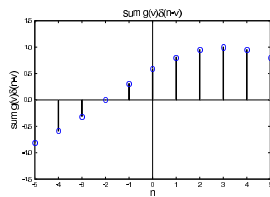


- We can now add all these together...

Notes

Sifting *cont'd.*

- Adding all the delta values together we get



- which is a discrete (*sifted*) representation of the original signal, $g(n)$.
- This process can be represented by

$$x[n] = \dots + g(-5)\delta(n+5) + g(-4)\delta(n+4) + \dots + g(4)\delta(n-4) + g(5)\delta(n-5) + \dots$$
- where $[\cdot]$ signifies a discrete formulation. This can be shortened to

$$x[n] = \sum_{k=-\infty}^{\infty} g(k)\delta(n-k). \text{ For our case } x[n] = \sum_{k=-5}^5 g(k)\delta(n-k).$$

Notes

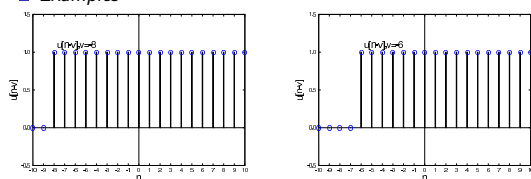
Unit Step Function

- The unit step function:

$$u[n-v] = \begin{cases} 1 & \text{if } n-v \geq 0, \\ 0 & \text{otherwise.} \end{cases}$$

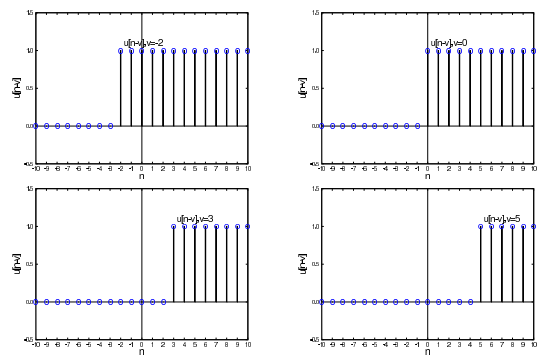
- switches from zero to unit value.

- *Examples*



Notes

Unit Step Function *cont'd.*



Notes

Unit Step Function

- It can be defined using the unit impulse function ($\delta[n-v]$):

$$u[n-v] = \sum_{m=v}^{\infty} \delta[n-m]$$

- Also

$$\delta[n-v] = u[n-v] - u[n-1-v].$$

- These are known as *recurrence* formula, where the current signal value is dependent on previous signal values:
“to recur”

Meaning: to repeat.

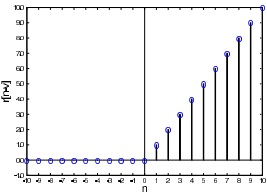
Notes

Ramp Function

- Another interesting function type is the ramp function.
- Given by

$$r[n-v] = (n-v)u[n].$$

Example



Notes

Digital Sinusoidal Functions

- Digital sine wave:

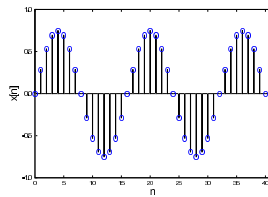
$$x[n] = a \sin(n\Omega + \theta)$$

- Digital cosine wave:

$$x[n] = a \cos(n\Omega + \theta)$$

- Ω is the *digital "frequency"* measured in *radians*
- 1 cycle every N samples. Also $\Omega = 2\pi/N$ so that $N = 2\pi/\Omega$

- **Example** $a = 0.75$, $\theta = 0$ and $\Omega = \pi/8$, therefore $N = 2 \times 8 = 16$ *i.e.* $x[n] = 0.75 \sin(n\pi/8)$:



Notes

Comparison with Analog Sine Function

- Compare to a continuous analog sine wave:

$$x(t) = a \sin(t\omega + \theta)$$

where t could be time in seconds and $\omega = 2\pi f$ is the angular frequency, therefore in *radians per second*.

- The interval between each sample n is T_s seconds, so there is a sample at every $t = nT_s$ seconds
- The continuous sine wave can then be written as

$$x(n) = a \sin(nT_s 2\pi f + \theta)$$

- If we equate the continuous and digital versions, then

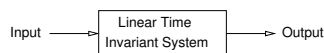
$$x[n] = x(n)$$

$$a \sin(n\Omega + \theta) = a \sin(nT_s 2\pi f + \theta)$$

- Therefore $\Omega = T_s 2\pi f$ or if sampling frequency is $f_s = 1/T_s$ then $\Omega = 2\pi f/f_s$.

Notes

Linear Time Invariant Systems



- **Time Invariance:**

- The same response to the same input at any time.

If $q[n - v_1] = q[n - v_2] = x[n]$ for constants v_1 and v_2 then

$$F(q[n - v_1]) = F(q[n - v_2])$$

- **Linear System:**

- *Principle of Superposition:*

- If the input consists of a sum of signals then the output is the sum of the responses to those signals, e.g.

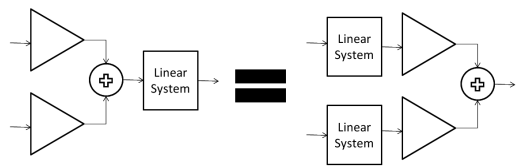
If the output of a system is $y_1[n]$ and $y_2[n]$ in response to two different inputs $x_1[n]$ and $x_2[n]$ respectively then the output of the same system for the two inputs weighted and combined *i.e.* $ax_1[n] + bx_2[n]$ will be $ay_1[n] + by_2[n]$ where a and b are constants.

Notes

Linear Time Invariant Systems

For a linear system $y[n] = F(x[n])$

$$F(ax_1[n] + bx_2[n]) = aF(x_1[n]) + bF(x_2[n])$$

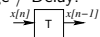


Notes

Linear Time Invariant Systems

■ A Linear Time Invariant (LTI) system consists of:

■ Storage / Delay:



■ Addition / Subtraction: e.g. $y[n] = x[n] + x[n - 1]$



■ Multiplication by Constants: e.g. $y[n] = \frac{1}{3}x[n]$

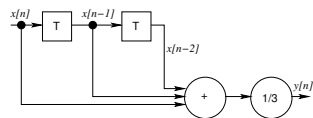


Notes

Simple LTI System Example

■ Example

Moving average filter, $y[n] = \frac{x[n] + x[n-1] + x[n-2]}{3}$



Notes

Other System Properties

An LTI system is

- Associative, where a system can be broken down into simpler subsystems for analysis or synthesis
- Commutative, where if a system is composed of a series of subsystems then the subsystems can be arranged in any order

LTI systems may also have

- Causality: output does not depend on future input values
- Stability: output is bounded for a bounded input (see Lecture 04)
- Invertibility: input can be uniquely found from the input (e.g. the square of a number is not invertible)
- Memory: output depends on past input values

Notes

Examples of Linear Mathematical Operations

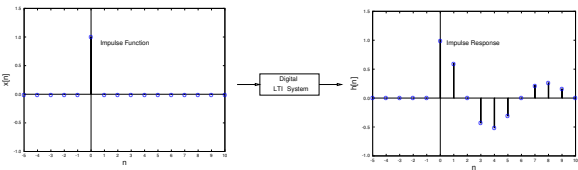
- Scaling (i.e. idealised gain or attenuation)
- Differentiation
- Integration
- The Laplace transform
- The Fourier transform
- The z-transform

Notes

Impulse Response

An LTI system possesses an Impulse Response which characterizes the system's output if an impulse function is applied to the input.

Example Impulse Response



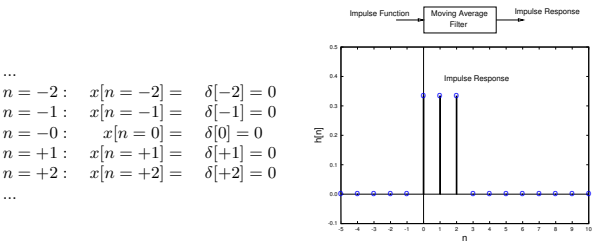
Notes

Impulse Response Example

Remember the moving average filter:

y[n] = 1/3 (x[n] + x[n-1] + x[n-2])

If the input is the impulse function: x[n = 0] = delta(0), then y[n] is the output in response to an impulse function, i.e. the impulse response hence h[n] = y[n]...



Notes

Impulse Response Example

Moving Average Filter:

y[n] = 1/3 (x[n] + x[n-1] + x[n-2])

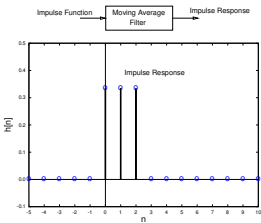
Example, Input Signal = Impulse Function:

x[n] = delta[n] = { 1 if n = 0, 0 elsewhere.

Therefore

- h[n < 0] = y[n < 0] = 0
- h[0] = y[0] = 1/3 (delta[0] + delta[-1] + delta[-2]) = 1/3
- h[1] = y[1] = 1/3 (delta[+1] + delta[0] + delta[-1]) = 1/3
- h[2] = y[2] = 1/3 (delta[+2] + delta[+1] + delta[0]) = 1/3
- h[n > 2] = y[n > 2] = 0

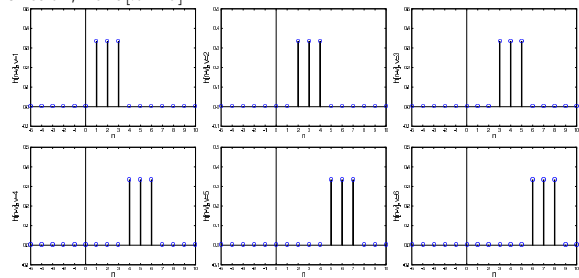
Response h[n] is known as the Impulse Response.



Notes

Impulse Response Examples - shifting

The impulse response can also be determined for a shifted impulse function, i.e. delta[n - v]



What will the system output (y[n]) be if the input consists of more than one impulse function shifted by different amounts?

Notes

System Response to Multiple Shifted Impulse Responses

What will the system output ($y[n]$) be if the input consists of more than one impulse function shifted by different amounts?

Remember that all LTI systems obey the “*Principle of Superposition*”...

So, for the inputs

$$x_1[n] = a\delta[n] \text{ and } x_2[n] = b\delta[n-1],$$

where a and b are constants, the corresponding outputs will be

$$y_1[n] = ah[n] \text{ and } y_2[n] = bh[n-1],$$

i.e. impulse responses. Therefore if

$$x[n] = x_1[n] + x_2[n] = a\delta[n] + b\delta[n-1] \text{ then}$$

$$y[n] = ah[n] + bh[n-1].$$

Notes

Other Signals: Step Function

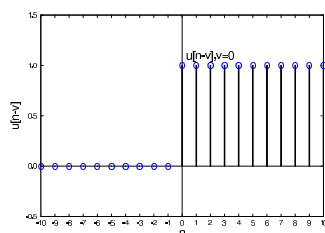
- The discrete step function can be thought of as a series of impulse functions (remember sifting).
- Each impulse function creates an impulse response.
- The output is then the joint response of all the impulse responses scaled by the inputs.

- A discretely sampled step input (starting at $n = 0$) is given by:

$$x[n] = \sum_{k=0}^{\infty} \delta[n-k].$$

- Therefore, using the *Principle of Superposition* we get

$$y[n] = \sum_{k=0}^{\infty} h[n-k].$$



Notes

Step Function Moving Average

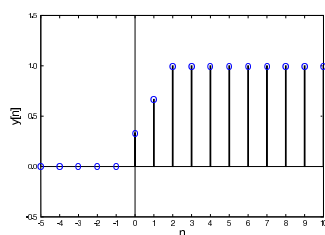
Moving average (with $k = 3$) has an impulse response:

- $h[n < 0] = y[n < 0] = 0$
- $h[0] = y[0] = \frac{1}{3}(\delta[0] + \delta[-1] + \delta[-2]) = \frac{1}{3}$
- $h[1] = y[1] = \frac{1}{3}(\delta[+1] + \delta[0] + \delta[-1]) = \frac{1}{3}$
- $h[2] = y[2] = \frac{1}{3}(\delta[+2] + \delta[+1] + \delta[0]) = \frac{1}{3}$
- $h[n > 2] = y[n > 2] = 0$

Moving average of a step function is then:

$$y[n] = \sum_{k=0}^{\infty} h[n-k]$$

$$= \begin{cases} 0 & \text{if } n \leq -1 \\ 1/3 & \text{if } n = 0 \\ 2/3 & \text{if } n = 1 \\ 1 & \text{if } n \geq 2 \end{cases}$$



Notes

Scaled Impulse Function Inputs

What happens when the step function is given by:

u[n-v] = { a if n-v >= 0, 0 otherwise }

The discrete impulse function

x[n] = sum_{k=0}^inf a delta[n-k]

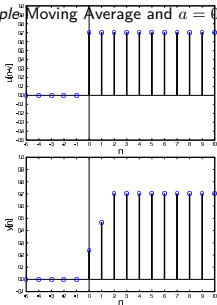
Using the Principle of Superposition:

y[n] = sum_{k=0}^inf ah[n-k]

Example Moving average filter, k = 3

y[n] = { 0 if n <= 0, a/3 if n = 0, 2a/3 if n = 1, a if n >= 2 }

Example Moving Average and a = 0.7



Notes

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Digital Convolution

What happens if the scale of the input impulse functions (a) varies with n? i.e.

x[n] = a[n]delta[n-k]

Using the Principle of Superposition we get

y[n] = sum_{k=-inf}^inf a[k]h[n-k]

This is known as the Convolution Sum.

Example

x[n] = { 0 if n < 0, a[0] if n = 0, a[1] if n = 1, 0 if n >= 2 }

which is the same as x[n] = a[0]delta[n] + a[1]delta[n-1]. Then

y[n] = a[0]h[n] + a[1]h[n-1]

Notes

Notes section with 8 horizontal lines for writing.

Digital Convolution Example

Q. Find y[n] if a[0] = 1 and a[1] = 2 using the impulse response of the moving average filter, k = 3.

A.

y[n] = a[0]h[n] + a[1]h[n-1] = h[n] + 2h[n-1]
y[-1] = h[-1] + 2h[-2] = 0 + 0 = 0
y[0] = h[0] + 2h[-1] = 1/3 + 0 = 1/3
y[1] = h[1] + 2h[0] = 1/3 + 2/3 = 1
y[2] = h[2] + 2h[1] = 1/3 + 2/3 = 1
y[3] = h[3] + 2h[2] = 0 + 2/3 = 2/3
y[4] = h[4] + 2h[3] = 0 + 0 = 0

Notes

Notes section with 8 horizontal lines for writing.

Digital Convolution Trivia

Convolution is often represented by an asterik:

y[n] = \sum_{k=-\infty}^{\infty} a[k]h[n-k] = a[n] * h[n]

Convolution is commutative:

y[n] = a[n] * h[n] = h[n] * a[n]

= \sum_{k=-\infty}^{\infty} h[k]a[n-k].

Convolution is associative: cascaded systems

{x[n] * h_1[n]} * h_2[n] = x[n] * {h_1[n] * h_2[n]}

Convolution is distributive: systems in parallel

x[n] * {h_1[n] + h_2[n]} = x[n] * h_1[n] + x[n] * h_2[n]

Notes

Digital Convolution Example

Original signal: (1 cycle every 104 samples)

x_1[n] = \sin(n\pi/52)

Noise signal: (1 cycle every 4 samples)

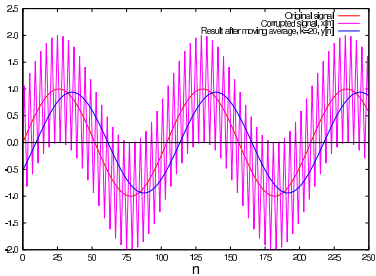
x_2[n] = \sin(n\pi/2).

Input signal:

x[n] = x_1[n] + x_2[n].

Moving average filter, k = 20:

y[n] = \frac{1}{20} \sum_{k=0}^{k=19} x[n-k].



Notes

Digital Cross-Correlation

Cross-correlation can be used to compare 2 signals.

- If $x_1[n]$ and $x_2[n]$ are two signals then digital cross-correlation is defined:

y[n] = \sum_{m=-\infty}^{\infty} x_1^*[m]x_2[n+m]

where $x_1^*[n]$ is the complex conjugate of $x_1[n]$.

- For a real signal $x_1^*[n] = x_1[n]$.
- l is the lag.
- If $x_1[n]$ and $x_2[n]$ are the same signal but with a delay between them, then $y[l]$ is at a maximum when l is equal to this delay.

Notes

Digital Cross-Correlation Example

Q. Given $x_1 = (0\ 0\ 0.5\ 0.7\ 0)^T$ and $x_2 = (0\ 0.5\ 0.7\ 0\ 0)^T$. Calculate the cross-correlation for these two real signals.

A. Cross correlation for a real signal is:

y[n] = sum_{m=-inf}^{inf} x1[m]x2[n+m]

There are 5 elements in these vectors so (changing the limits):

y[n] = sum_{m=0}^4 x1[m]x2[n+m]

We can then calculate the results. Some example calculations:

y[l = 0] = x1[0] x x2[0] + x1[1] x x2[1] + x1[2] x x2[2] + x1[3] x x2[3] + x1[4] x x2[4] = 0 x 0 + 0 x 0.5 + 0.5 x 0.7 + 0.7 x 0 + 0 x 0 = 0.5 x 0.7 = 0.35

y[l = 1] = x1[0] x x2[0 + 1] + x1[1] x x2[1 + 1] + x1[2] x x2[2 + 1] + x1[3] x x2[3 + 1] + x1[4] x x2[4 + 1] = 0 x 0.5 + 0 x 0.7 + 0 x 0 + 0 x 0 + 0 x 0 = 0

Notes

Digital Cross-Correlation cont'd.

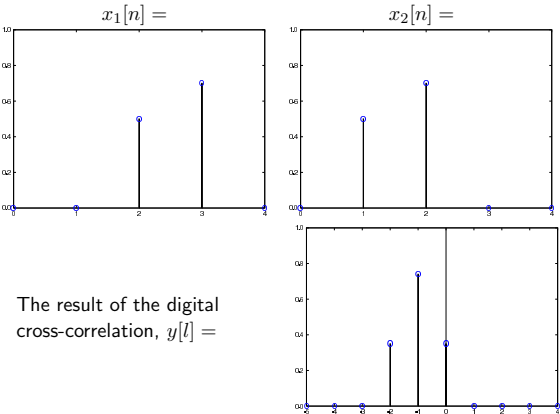
Here are the results for each combination of l and m values:

l	m					y[l]
	0	1	2	3	4	
-5	0	0	0	0	0	0
-4	0	0	0	0	0	0
-3	0	0	0	0	0	0
-2	0	0	0	0.35	0	0.35
-1	0	0	0.25	0.49	0	0.74
0	0	0	0.35	0	0	0.35
1	0	0	0	0	0	0
2	0	0	0	0	0	0
3	0	0	0	0	0	0
4	0	0	0	0	0	0

- A peak at $l = -1$.
- l is the lag, so there is a lag of -1 .
- This means x_1 has some similar signal as x_2 but lagged by 1 step.
- We can also see from the signal definitions $x_1 = (0\ 0\ 0.5\ 0.7\ 0)^T$ and $x_2 = (0\ 0.5\ 0.7\ 0\ 0)^T$ that $x_1[n-1] = x_2[n]$.

Notes

Digital Cross-Correlation cont'd.



The result of the digital cross-correlation, $y[l] =$

Notes

Difference Equations

Difference equations are the name given to the equations that describe the digital signals and systems. For example the equation for the moving average filter with $k = 3$:

$$y[n] = \frac{x[n] + x[n - 1] + x[n - 2]}{3}$$

is known as a difference equation.
Difference equations for LTI systems can always be put in the form:

$$\sum_{m=0}^N a[m]y[n - m] = \sum_{m=0}^M b[m]x[n - m].$$

So for our moving average filter:

- $M = 2$ and $N = 0$.
- $a[m]$ and $b[m]$ are known as coefficients.
- For the moving average output y there is only one coefficient, $a[0] = 1$.
- For the moving average input x , there are three coefficients $b[0] = b[1] = b[2] = \frac{1}{3}$.

Notes

Summary

Today we have covered

- Types of digital signal, e.g. unit impulse function
- Sifting
- Digital sine and cosine functions
- Linear time invariant (LTI) systems
- Impulse response
- Moving average of a step function
- Digital convolution
- Digital cross-correlation
- Generalized difference equation for LTI systems

Notes

Notes
