Discrete-time Signals and Systems Summary

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Discrete-time Signals and Systems



- Discrete-time Signals
- Discrete-time Systems
- 1D Convolutions
- FIR and IIR Systems



Discrete-time Signals



A discrete-time signal is mathematically represented as a sequence of data samples (typically real numbers), in which the n-th sequence sample is denoted by x(n):

$$\{x(n)\}, -\infty < n < \infty,$$

where n is an integer index.

A discrete-time signal x(n) is a function of the form:

x(n): $\mathbb{Z} \to \mathbb{R}$.

Complex signal: x(n): $\mathbb{Z} \to \mathbb{C}$.



Discrete-time Signals



A digital signal x(n) is a discrete-time function of the form: x(n): $\mathbb{Z} \to [0, ..., 2^B - 1],$

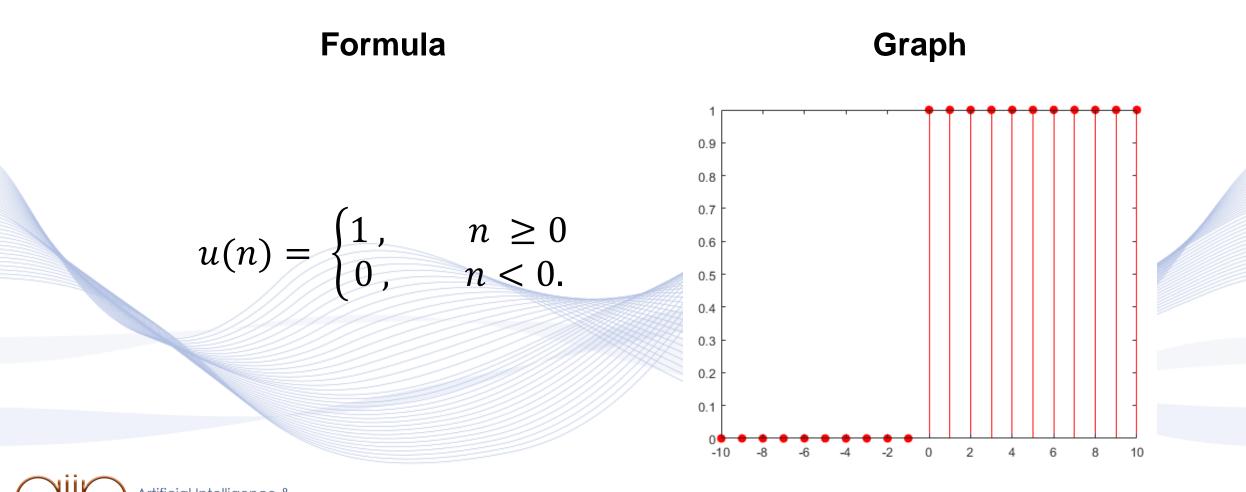
when signal sample has B bits per sample.

• If B = 8, each signal sample has 256 distinct values.





Unit step signal



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Complex Exponential Signal

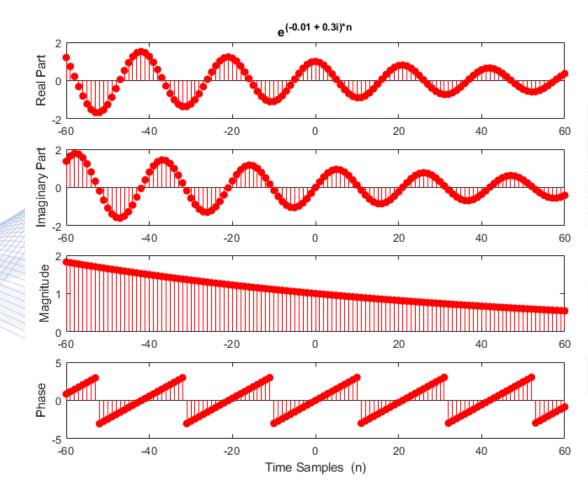
Formula

Graph

 $x(n) = a e^{i\omega n} = a \cos(\omega n) + i a \sin(\omega n).$

- a: complex signal magnitude.
- $\varphi(\omega) = \omega n$: complex signal **phase**.







y(n)

Discrete-time Systems

A *discrete-time system* is defined mathematically as a transformation that maps an input sequence x(n) into an output sequence y(n):

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

T[x(n)]



x(n)

VML

1D Linear Systems

Linear Shift-Invariant (LSI) systems:

• Linearity:

$$T[ax_1 + bx_2] = aT[x_1] + bT[x_2].$$

• Shift-Invariance:

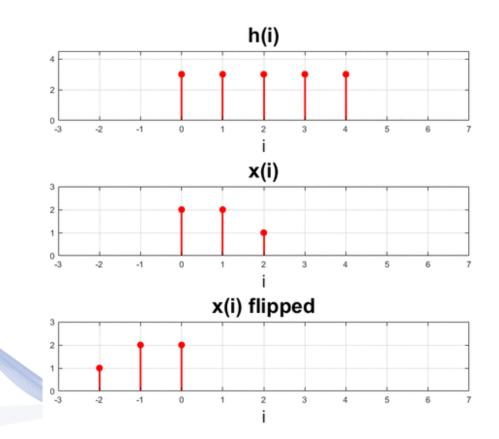
$$y(n) = T[x(n)] \Rightarrow y(n-m) = T[x(n-m)].$$

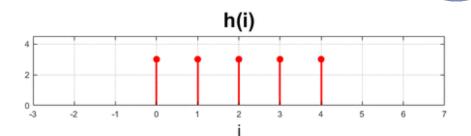
 LSI system is defined by its *impulse response function* h(n) and *convolution* operator:

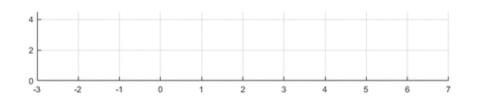
y(n) = h(n) * x(n).

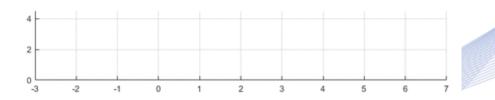


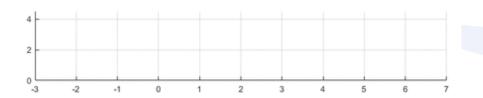
Linear 1D convolution – Example **(VML**







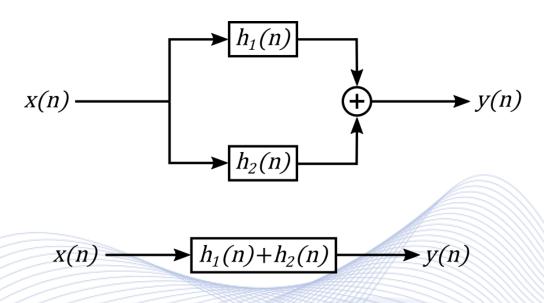




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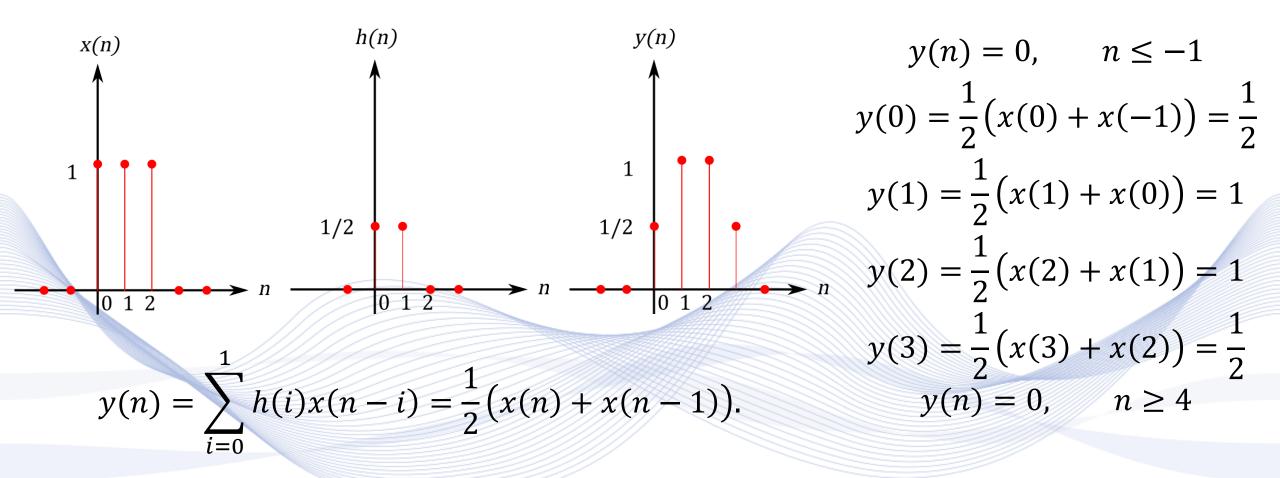
Convolution Properties



LTI System distributivity.

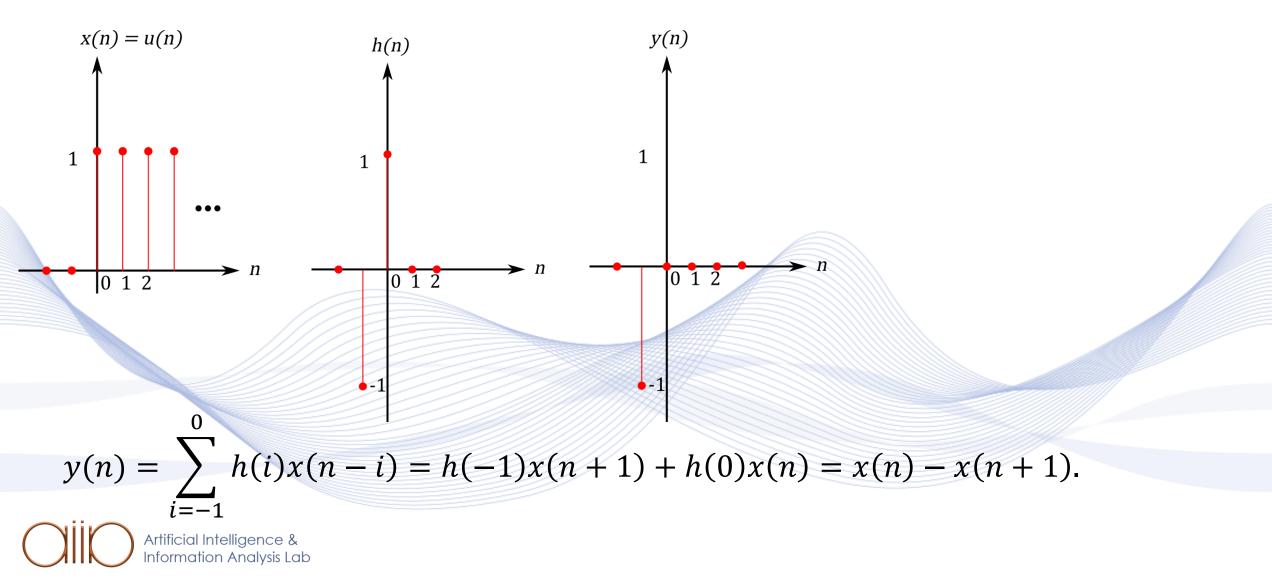












Calculate the output of a LTI system whose impulse response is:

ML

$$h(n) = \begin{cases} 1, & 0 \le n \le M - 1\\ 0, & n < 0, & n > M - 1. \end{cases}$$

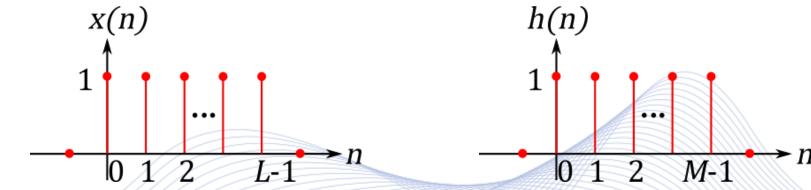
while the input is:
$$x(n) = \begin{cases} 1, & 0 \le n \le L - 1\\ 0, & n < 0, & n > L - 1. \end{cases}$$

where $M < L$.





- We have to convolute two rectangular pulses.
- Input duration is *L*, while impulse response duration is *M* samples.



- It is given that M < L.
- We can conclude that the output will be a trapezoid because

Linear 1D correlation



• Correlation of template signal h and input signal x(n) (inner product):

$$r_{hx}(n) = \sum_{i=0}^{M-1} h(i)x(n+i) = \mathbf{h}^T \mathbf{x}(n).$$

- $\mathbf{h} = [h(0), ..., h(M 1)]^T$: template vector.
- $\mathbf{x}(n) = [x(n), ..., x(n + M 1)]^T$: local signal vector.



2D convolutions



• A 2D linear shift invariant system is described by a 2D convolution of input *x* with a convolutional kernel *h*:

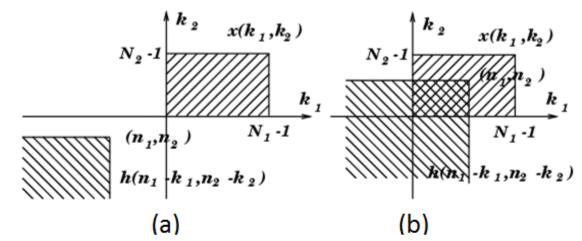
$$y(n_1, n_2) = h(n_1, n_2) * x(n_1, n_2) = \sum_{i_1} \sum_{i_2} h(i_1, i_2) x(n_1 - i_1, n_2 - i_2).$$

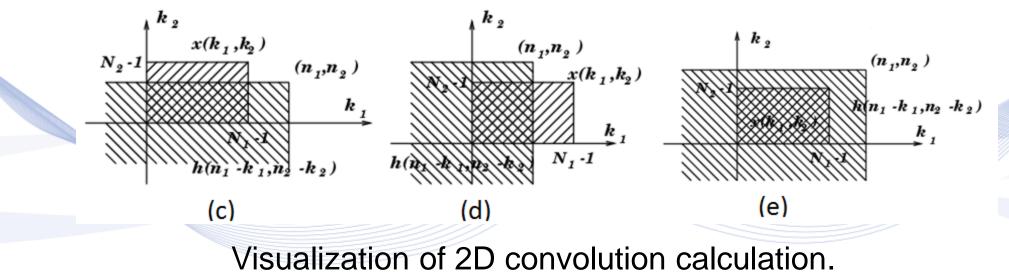
- Input x has typically limited region of support (size), e.g., it can be an image of $M_1 \times M_2$ pixels.
- Convolutional kernel h may have limited or infinite region of support.





2D convolutions





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FIR and IIR Systems

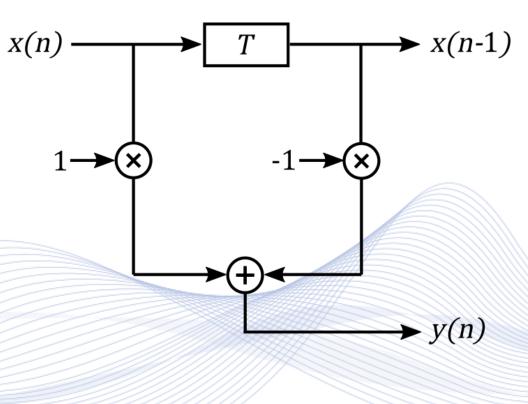
If a system has an impulse response of finite length h(i), i = 1, ..., M, it is called *Finite Impulse Response (FIR)* system:

$$y(n) = h(n) * x(n) = \sum_{i=0}^{M-1} h(i)x(n-i).$$





Numerical differentiator

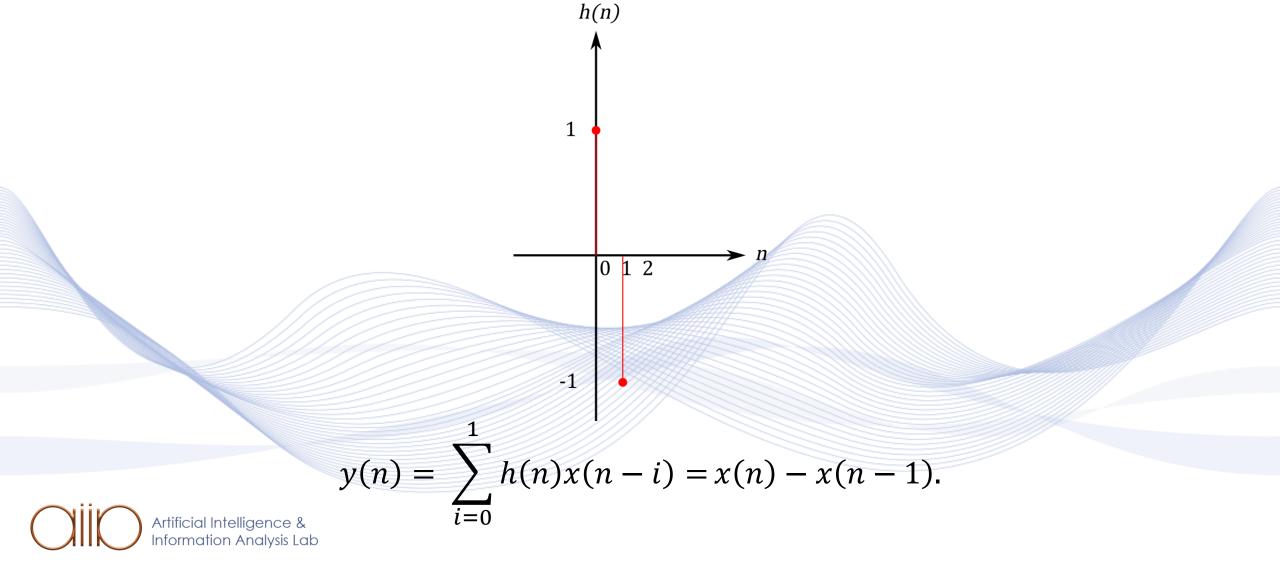


y(n) = x(n) - x(n-1).





Numerical differentiator







If a system has an impulse response of infinite length $h(i), i = -\infty, ..., \infty$, is referred as an *Infinite Impulse Response (IIR)* system.

An IIR system is described by a *difference equation*:

$$\sum_{k=0}^{N} a_k y(n-k) = \sum_{k=0}^{M} b_k x(n-k),$$

having coefficients a_k , k = 0, ..., N and b_k , k = 0, ... M.





IIR Systems

An IIR system can also be described by a *recursive* formula:

$$y(n) = \frac{1}{a_0} \left(\sum_{i=0}^{M} b_i x(n-i) - \sum_{i=1}^{N} a_i y(n-i) \right)$$

 Output y(n) is given at time n in terms of the present input sample and the previous values of the input and output samples.

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