

Digital Signal Processing

II. Discrete signals and systems

II.1 Discrete signals

Representation

A discrete signal can be represented:

- ▶ graphically
- ▶ in table form
- ▶ as a vector: $x[n] = [..., 0, 0, 1, 3, 4, 5, 0, ...]$
 - ▶ an **arrow** indicates the origin of time ($n = 0$).
 - ▶ if the arrow is missing, the origin of time is at the first element
 - ▶ the dots ... indicate that the value remains the same from that point onwards

Examples: at blackboard

Notation: $x[4]$ represents the value of the fourth sample in the signal $x[n]$

Basic signals

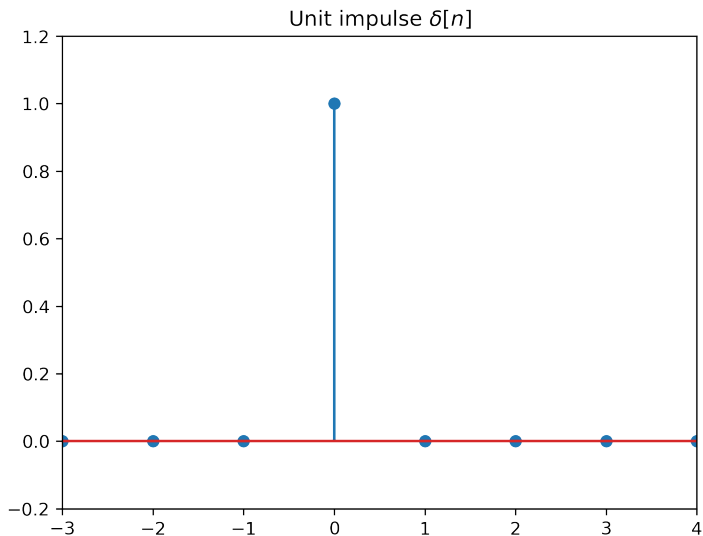
Some elementary signals are presented below.

Unit impulse

Contains a single non-zero value of 1 located at time 0. It is denoted with $\delta[n]$.

$$\delta[n] = \begin{cases} 1 & \text{if } n = 0 \\ 0 & \text{otherwise} \end{cases}.$$

Representation

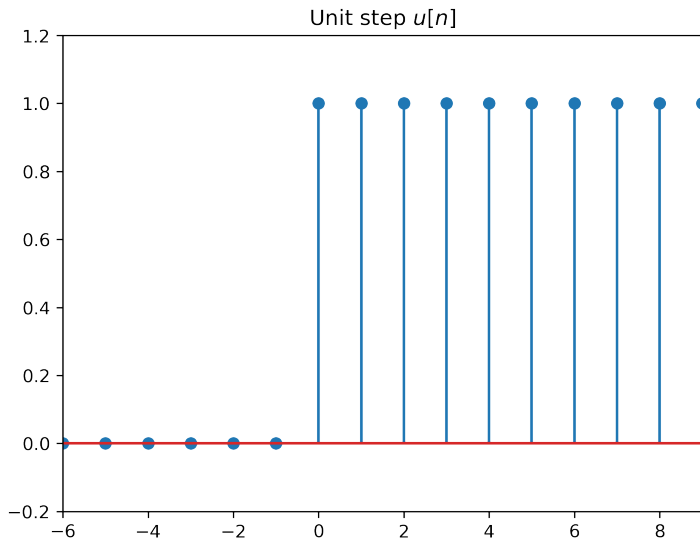


Unit step

It is denoted with $u[n]$.

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

Representation

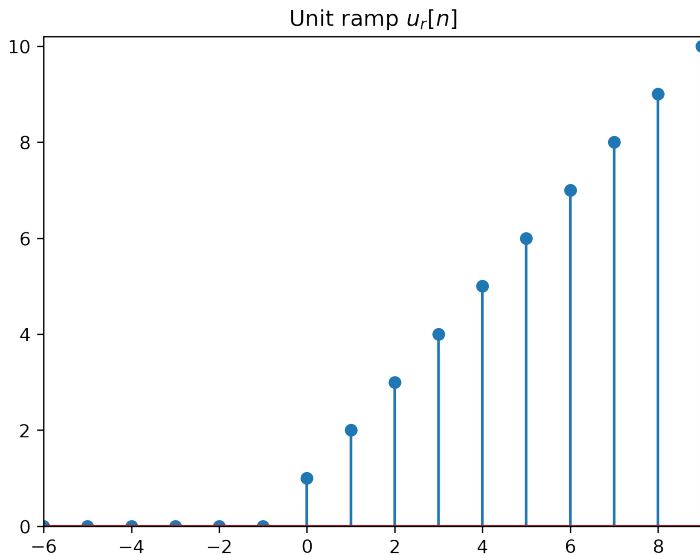


Unit ramp

It is denoted with $u_r[n]$.

$$u_r[n] = \begin{cases} n & \text{if } n \geq 0 \\ 0 & \text{otherwise} \end{cases}.$$

Representation



Exponential signal

Exponential signal

It does not have a special notation. It is defined by:

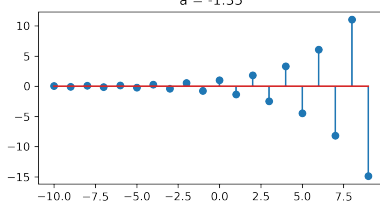
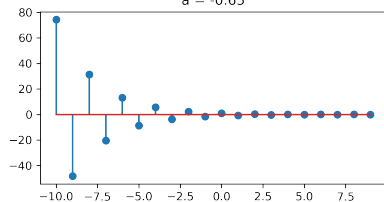
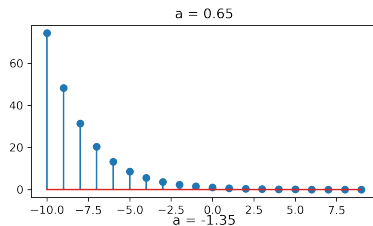
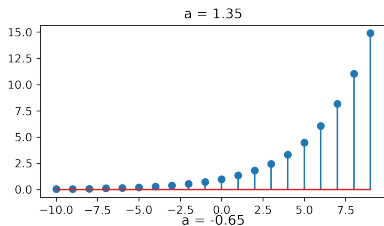
$$x[n] = a^n.$$

a can be a real or a complex number. Here we consider only the case when a is real.

Depending on the value of a , we have four possible cases:

1. $a \geq 1$
2. $0 \leq a < 1$
3. $-1 < a < 0$
4. $a \leq -1$

Representation



Discrete-time sinusoids

- ▶ A real sinusoid is defined as

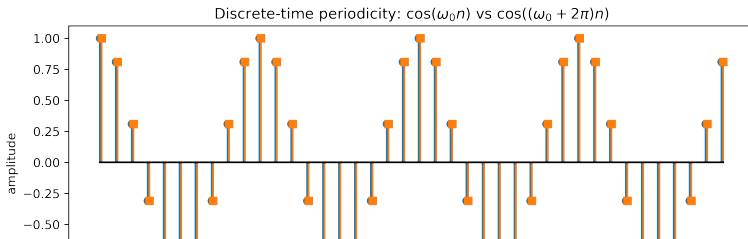
$$x[n] = A \cos(\omega_0 n + \varphi),$$

where ω_0 is the normalized rad/sample frequency and φ the phase.

- ▶ Complex exponentials and sinusoids are tightly related:

$$e^{j\omega_0 n} = \cos(\omega_0 n) + j \sin(\omega_0 n).$$

- ▶ Periodicity in discrete time: $e^{j(\omega_0 + 2\pi k)n} = e^{j\omega_0 n}$, thus frequencies differing by multiples of 2π are indistinguishable. The sinusoid is periodic iff there exist integers N_0, m such that $\omega_0 N_0 = 2\pi m$.



II.2 Types of discrete signals

Signals with finite energy

- ▶ The **energy of a discrete signal** is defined as

$$E = \sum_{n=-\infty}^{\infty} |x[n]|^2.$$

- ▶ If E is finite, the signal is said to have finite energy.
- ▶ Examples:
 - ▶ unit impulse has finite energy
 - ▶ unit step does not

Connection with DEDP class

- ▶ Cross-link with DEDP course:

$$E = \|\mathbf{x} - \mathbf{0}\|^2 = \|\mathbf{x}\|^2$$

- ▶ Energy of a signal = **squared Euclidean distance to 0**
 - ▶ geometric interpretation: squared length of the segment from 0 to the point \mathbf{x}
 - ▶ holds for continuous signals as well:

$$E = \|\mathbf{x}\|^2 = \int_{-\infty}^{\infty} |x(t)|^2 dt$$

Signals with finite power

- ▶ The **average power of a discrete signal** is defined as

$$P = \lim_{N \rightarrow \infty} \frac{\sum_{n=-N}^N |x[n]|^2}{2N + 1}.$$

- ▶ In other words, the average power is the average energy per sample.
- ▶ If P is finite, the signal is said to have finite power.
- ▶ Every finite-energy (square-summable) signal has zero average power, $P = 0$. A signal with infinite energy can have finite (e.g., periodic signals) or infinite power.
- ▶ Example: unit step has finite power $P = \frac{1}{2}$ under the symmetric averaging definition (proof at blackboard).

Power/energy examples

- ▶ $\delta[n]$: $E = 1$, $P = 0$.
- ▶ $\cos(\omega_0 n)$: $E = \infty$, $P = \frac{1}{2}$.

Estimated $P[\text{delta}] \sim 0.000$, $P[\text{cos}] \sim 0.500$

Connection with DEDP class

- ▶ Average power = temporal average squared value $\overline{X^2}$
 - ▶ i.e. average value of the square of samples

Periodic and non-periodic signals

- ▶ A signal is called **periodic** if its values repeat after N samples (the **period**):

$$x[n] = x[n + N], \forall n.$$

- ▶ The **fundamental period** of a signal is the minimum value of N .
- ▶ Periodic signals have infinite energy, and finite power equal to the power of a single period.

Even and odd signals

- ▶ A real signal is **even** if it satisfies the following symmetry:

$$x[n] = x[-n], \forall n.$$

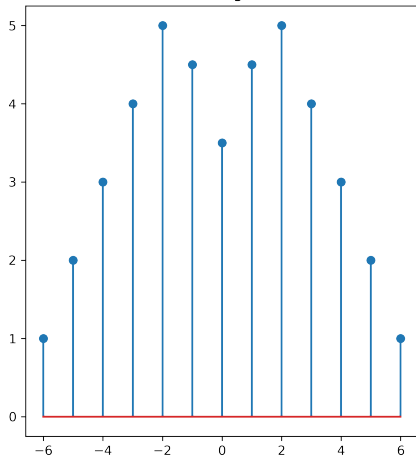
- ▶ A real signal is **odd** if it satisfies the following anti-symmetry:

$$-x[n] = x[-n], \forall n.$$

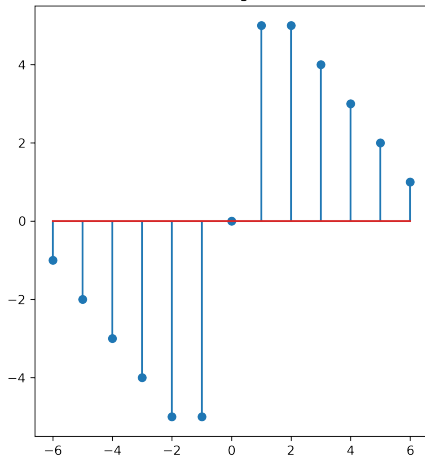
- ▶ There exist signals which are neither even nor odd.

Even and odd signals: example

Even signal



Odd signal



Even and odd parts of a signal

- ▶ Every signal can be written as the sum of an even signal and an odd signal:

$$x[n] = x_e[n] + x_o[n]$$

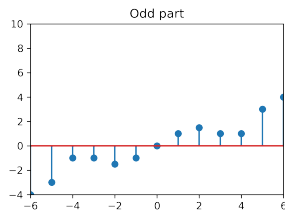
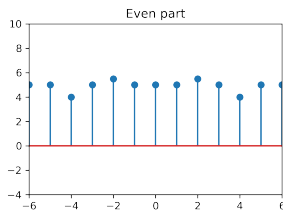
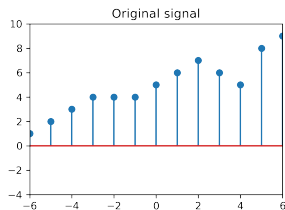
- ▶ The even and the odd parts of the signal can be found as follows:

$$x_e[n] = \frac{x[n] + x[-n]}{2}.$$

$$x_o[n] = \frac{x[n] - x[-n]}{2}.$$

- ▶ Proof: check that $x_e[n]$ is even, $x_o[n]$ is odd, and their sum is $x[n]$

Even and odd parts: example

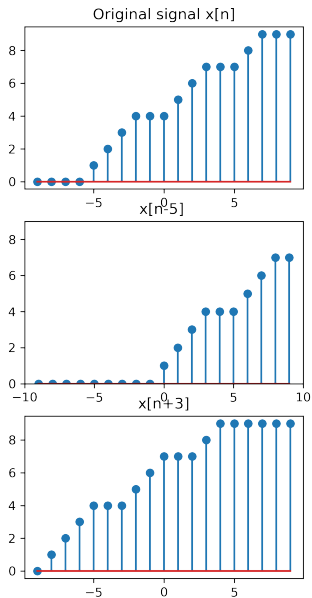


II.3 Basic operations with discrete signals

Time shifting

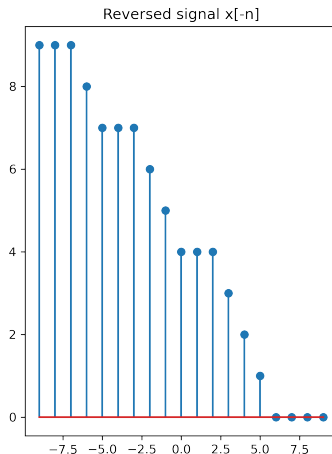
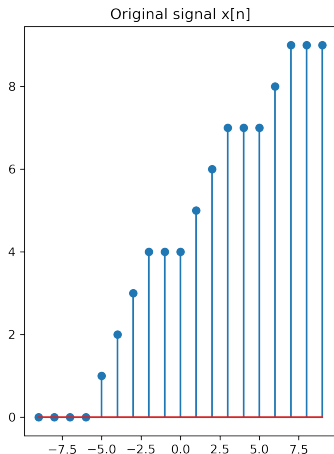
- ▶ The signal $x[n - k]$ is $x[n]$ **delayed with k time units**
 - ▶ Graphically, $x[n - k]$ is shifted k units to the **right** compared to the original
- ▶ The signal $x[n + k]$ is $x[n]$ **anticipated with k time units**
 - ▶ Graphically, $x[n + k]$ is shifted k units to the **left** compared to the original signal.

Time shifting: representation



Time reversal

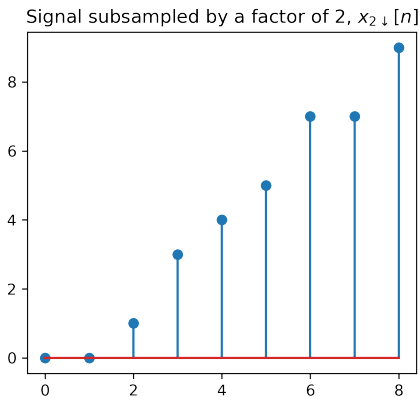
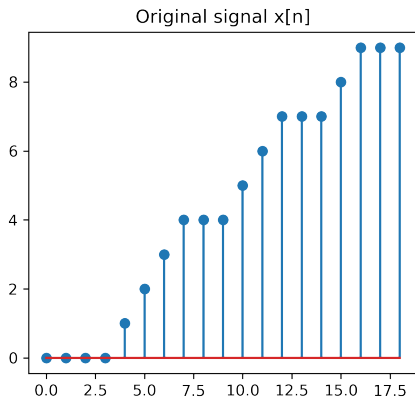
- Changing the variable n to $-n$ produces a signal $x[-n]$ which mirrors $x[n]$.



Subsampling

- **Subsampling** by a factor M : keep only one sample out of every M samples of the original signal.
 - Total number of samples is reduced M times

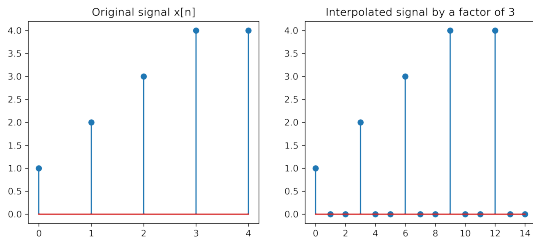
$$x_{M\downarrow}[n] = x[Mn]$$



Interpolation

- **Interpolation** by a factor of L adds $(L - 1)$ zeros between two samples in the original signal
 - Total number of samples increases L times

$$x_{L\uparrow} = \begin{cases} x[\frac{n}{L}] & \text{if } \frac{n}{L} \in \mathbb{N} \\ 0 & \text{otherwise} \end{cases}$$



- **Note:** Subsampling can introduce aliasing. In practice, apply an anti-alias low-pass filter before downsampling.

Mathematical operations

- ▶ A signal $x[n]$ can be **scaled** by a constant A , i.e. each sample is multiplied by A :

$$y[n] = Ax[n].$$

- ▶ Two signals $x_1[n]$ and $x_2[n]$ can be **summed** by summing the individual samples:

$$y[n] = x_1[n] + x_2[n]$$

- ▶ Two signals $x_1[n]$ and $x_2[n]$ can be **multiplied** by multiplying the individual samples:

$$y[n] = x_1[n] \cdot x_2[n]$$

II.4 Discrete systems

Definition

- ▶ **System** = a device or algorithm which produces an **output signal** based on an **input signal**
- ▶ We will only consider systems with a single input and a single output
- ▶ Figure here: blackboard.
- ▶ Common notation:
 - ▶ $x[n]$ is the input
 - ▶ $y[n]$ is the output
 - ▶ H is the system.

Notations

► Notations:

$$y[n] = H[x[n]]$$

(“the system H applied to the input $x[n]$ produces the output $y[n]$ ”)

$$x[n] \xrightarrow{H} y[n]$$

(“the input $x[n]$ is transformed by the system H into $y[n]$ ”)

Equations

- Usually, a system is described by the **input-output equation** (or **difference equation**) which explains how $y[n]$ is defined in terms of $x[n]$.

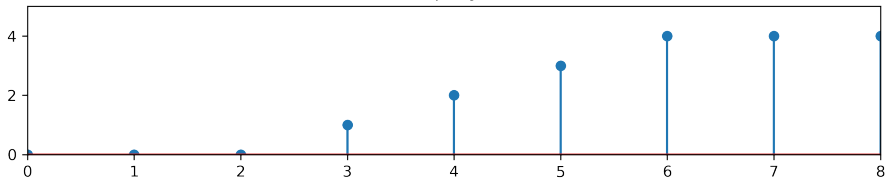
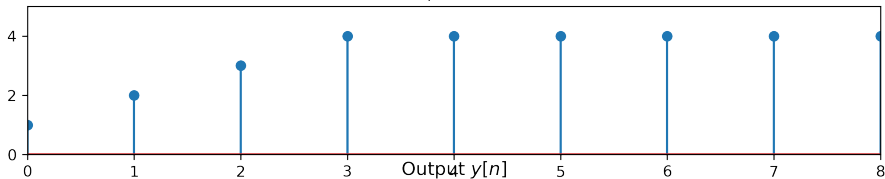
Examples:

1. $y[n] = x[n]$ (the identity system)
2. $y[n] = x[n - 3]$
3. $y[n] = x[n + 1]$
4. $y[n] = \frac{1}{3}(x[n + 1] + x[n] + x[n - 1])$
5. $y[n] = \max(x[n + 1], x[n], x[n - 1])$
6. $y[n] = (x[n])^2 + \log_{10} x[n - 1]$
7. $y[n] = \sum_{k=-\infty}^n x[k] = x[n] + x[n - 1] + x[n - 2] + \dots$

Example

$$y[n] = x[n - 3]$$

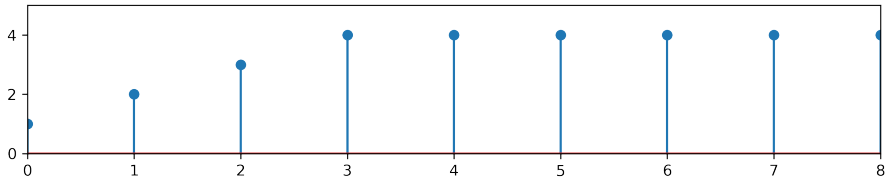
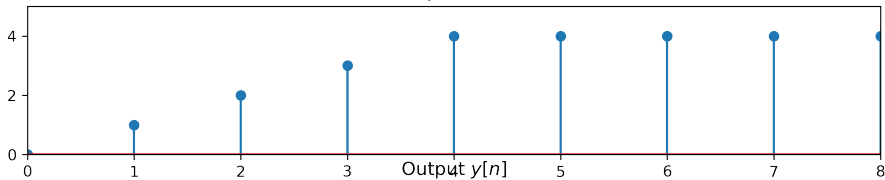
Input $x[n]$



Example

$$y[n] = x[n + 1]$$

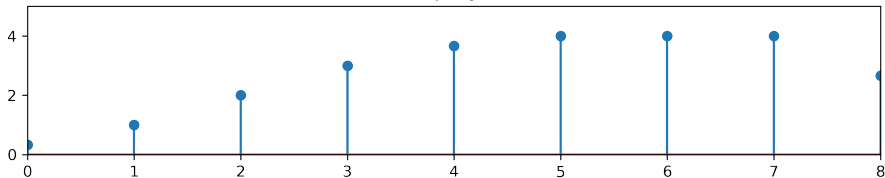
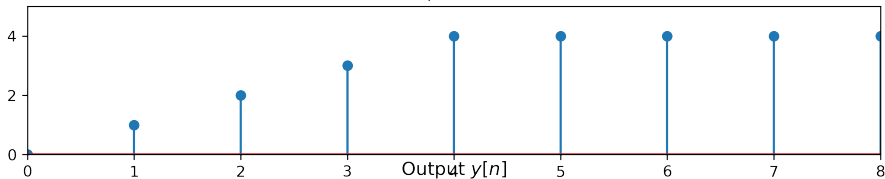
Input $x[n]$



Example

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

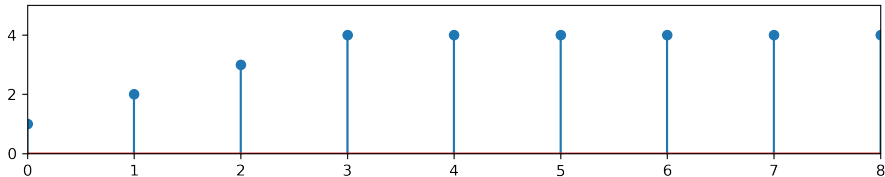
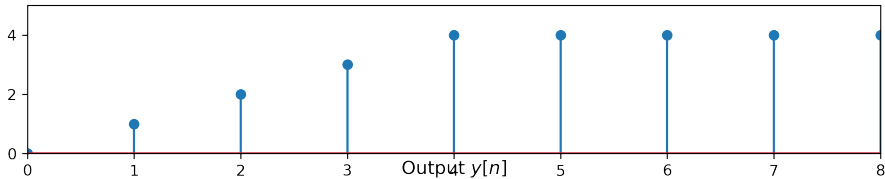
Input $x[n]$



Example

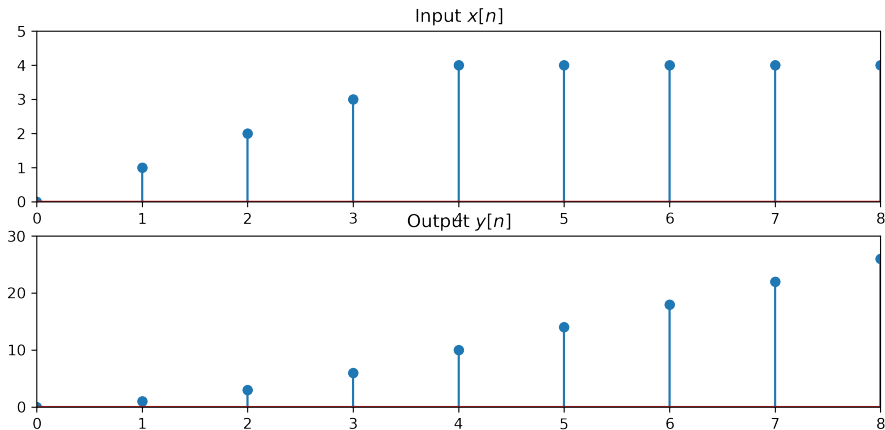
$$y[n] = \max(x[n+1], x[n], x[n-1])$$

Input $x[n]$



Example

$$y[n] = \sum_{k=-\infty}^n x[k] = x[n] + x[n-1] + x[n-2] + \dots$$



Recursive systems

- ▶ Some systems can/must be written in **recursive form**

$$y[n] = y[n - 1] + x[n],$$

- ▶ Must always specify **initial conditions**
 - ▶ i.e. initial value (e.g. $y[-1] = 2.5$)
 - ▶ if not mentioned, assume they are 0 (“relaxed system”)
 - ▶ they represent the internal state of the system at the starting moment
- ▶ For recursive systems, the output signal depends on both the input signal **and** on the initial conditions
 - ▶ different initial conditions lead to different outputs, even if input signal is the same
 - ▶ a recursive system with non-zero initial conditions can produce an output signal even in the absence of an input ($x[n] = 0$)

Representation of systems

- ▶ The operation of a system can be described graphically (see examples on blackboard):
 - ▶ summation of two signals
 - ▶ scaling of a signal with a constant
 - ▶ multiplication of two signals
 - ▶ delay element
 - ▶ anticipation element
 - ▶ other blocks for more complicated math operations

II.5 Classification of discrete systems

Memoryless / systems with memory

- ▶ **Memoryless (or static)**: output at time n depends only on the input **from the same moment** n
- ▶ Otherwise, the system **has memory (dynamic)**
- ▶ Examples:
 - ▶ memoryless: $y[n] = (x[n])^3 + 5$
 - ▶ with memory: $y[n] = (x[n])^3 + x[n - 1]$

Memoryless / systems with memory

- ▶ Memory of size N :
 - ▶ output at time n $y[n]$ depends only up to the last N inputs, $x[n - N], x[n - (N - 1)], \dots, x[n]$,
 - ▶ if N is finite: the system has **finite memory**
 - ▶ if $N = \infty$, the system has **infinite memory**
- ▶ Examples:
 - ▶ finite memory of order 4: $y[n] = x[n] + x[n - 2] + x[n - 4]$
 - ▶ infinite memory: $y[n] = 0.5y[n - 1] + 0.8x[n]$
 - ▶ recursive systems usually have infinite memory

Memoryless / systems with memory

- ▶ An input sample has an effect on the output only for the next N time moments
- ▶ For systems with infinite memory, any sample influences **all** the following samples, forever
 - ▶ but, if system is stable, the influence gets smaller and smaller

Time-Invariant and Time-Variant systems

- ▶ A relaxed system H is **time-invariant** if and only if:

$$x[n] \xrightarrow{H} y[n]$$

implies, $\forall x[n], \forall k$, that

$$x[n - k] \xrightarrow{H} y[n - k]$$

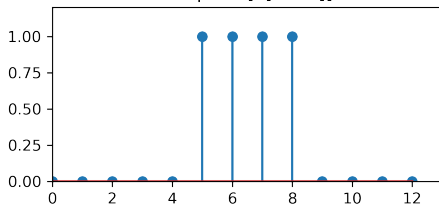
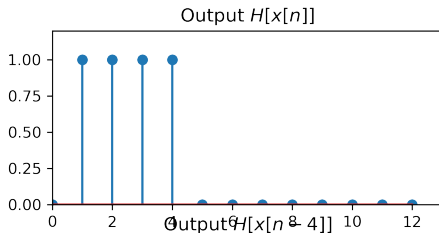
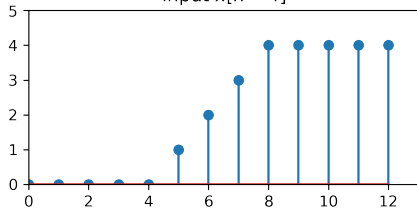
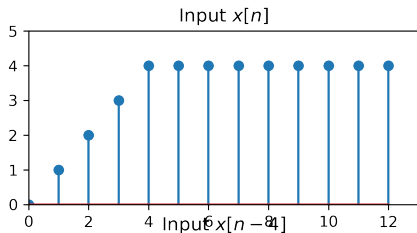
- ▶ Delaying the input signal with k will only delay the output with the same amount, otherwise the output is not affected
 - ▶ Must be true for all input signals, for all possible delays (positive or negative)
- ▶ Otherwise, the system is said to be **time-variant**

Time-Invariant and Time-Variant systems

- ▶ Examples:
 - ▶ $y[n] = x[n] - x[n - 1]$ is time-invariant
 - ▶ $y[n] = n \cdot x[n]$ is not time-invariant
- ▶ A system is time-invariant if it depends on n only through the input signal $x[n]$

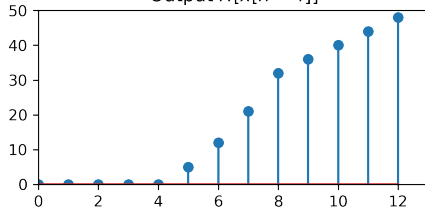
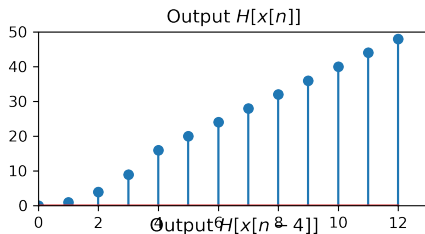
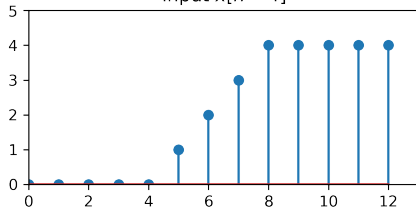
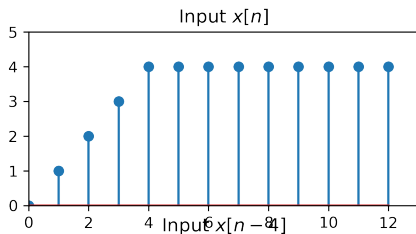
Example

Time-invariant system $y[n] = x[n] - x[n - 1]$



Another example

Time-variant system $y[n] = n \cdot x[n]$



Linear and nonlinear systems

- ▶ A system H is **linear** if:

$$H[ax_1[n] + bx_2[n]] = aH[x_1[n]] + bH[x_2[n]].$$

- ▶ Composed of two parts:
 - ▶ Applying the system to a sum of two signals = applying the system to each signal, and adding the results
 - ▶ Scaling the input signal with a constant a is the same as scaling the output signal with a
- ▶ The same relation will be true for a sum of many signals, not just two

Linear and nonlinear systems

- ▶ Advantage of linear systems
 - ▶ Complicated input signals can be decomposed into a sum of smaller parts
 - ▶ The system can be applied to each part independently
 - ▶ Then the results are added back
- ▶ Examples:
 - ▶ linear system: $y[n] = 3x[n] + 5x[n - 2]$
 - ▶ nonlinear system: $y[n] = 3(x[n])^2 + 5x[n - 2]$

Linear and nonlinear systems

- ▶ For a system to be linear, the input samples $x[n]$ must not undergo non-linear transformations.
- ▶ **The only transformations** of the input $x[n]$ allowed to take place in a linear system are:
 - ▶ scaling (multiplication) with a constant
 - ▶ delaying
 - ▶ summing different delayed versions of the signal (not summing with a constant)

Causal and non-causal systems

- ▶ **Causal:** the output $y[n]$ depends only on the current input $x[n]$ and the past values $x[n-1]$, $x[n-2]$, ..., but not on the future samples $x[n+1]$, $x[n+2]$, ...
- ▶ Otherwise the system is **non-causal**.
- ▶ A causal system can operate in real-time
 - ▶ we need only the input samples from the past
 - ▶ non-causal systems need samples from the future
- ▶ Examples:
 - ▶ $y[n] = x[n] - x[n-1]$ is causal
 - ▶ $y[n] = x[n+1] - x[n-1]$ is non-causal
 - ▶ $y[n] = x[-n]$ is non-causal

Stable and unstable systems

- ▶ **Bounded** signal: if there exists a value M such that all the samples of the signal are smaller than M in absolute value

$$x[n] \in [-M, M]$$

$$|x[n]| \leq M$$

- ▶ **Stable** system: if for any bounded input signal it produces a bounded output signal
 - ▶ not necessarily with the same M
 - ▶ known as BIBO (Bounded Input \rightarrow Bounded Output)
- ▶ In other words: when the input signal has bounded values, the output signal does not go towards ∞ or $-\infty$.

Stable and unstable systems

► Examples:

► $y[n] = (x[n])^3 - x[n + 4]$ is stable

► $y[n] = \frac{1}{x[n] - x[n-1]}$ is unstable

► $y[n] = \sum_{k=-\infty}^n x[k] = x[n] + x[n-1] + x[n-2] + \dots$ is unstable

Impulse response of Linear Time-Invariant (LTI) systems

Linear Time-Invariant (LTI) systems

- ▶ Notation: An **LTI** system (**L**inear **T**ime-**I**nv**I**variant) is a system which is simultaneously **linear** and **time-invariant**.
- ▶ LTI systems have an equation like this:

$$\begin{aligned}y[n] &= -a_1y[n-1] - a_2y[n-2] - \dots - a_Ny[n-N] + \\&\quad + b_0x[n] + b_1x[n-1] + \dots + b_Mx[n-M] \\&= -\sum_{k=1}^N a_ky[n-k] + \sum_{k=0}^M b_kx[n-k]\end{aligned}$$

- ▶ the above is for causal systems; non-causal can also have $[n+k]$

The impulse response

- ▶ **Impulse response** of a system = output (response) of when the input signal is the impulse $\delta[n]$:

$$h[n] = H(\delta[n])$$

- ▶ The impulse response of a LTI system **fully characterizes the system**:
 - ▶ based on $h[n]$ we can compute the response of the system to **any** input signal
 - ▶ all the properties of LTI systems can be described via characteristics of the impulse response

Signals are a sum of impulses

- ▶ Any signal $x[n]$ can be composed as **a sum of scaled and delayed impulses** $\delta[n]$.
- ▶ Example:
 $x[n] = \{3, 1, -5, 0, 2\} = 3\delta[n] + \delta[n-1] - 5\delta[n-2] + 0\cdot\delta[n-3] + 2\delta[n-4]$
- ▶ In general

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

i.e. a sum of impulses $\delta[n]$, each one delayed with k and scaled with the corresponding value $x[k]$

Convolution

- ▶ The response of a LTI system to a sum of impulses, delayed with k and scaled with $x[k]$, **is a sum of impulse responses, delayed with k and scaled with $x[k]$.**

$$\begin{aligned}y[n] &= H(x[n]) \\&= H\left(\sum_{k=-\infty}^{\infty} x[k]\delta[n-k]\right) \\&= \sum_{k=-\infty}^{\infty} x[k]H(\delta[n-k]) \\&= \sum_{k=-\infty}^{\infty} x[k]h[n-k].\end{aligned}$$

Convolution

- ▶ Convolution in short:
 - ▶ The input signal is composed of separate impulses
 - ▶ Each impulse will generate its own response (LTI)
 - ▶ Output signal is the sum of impulse responses, delayed and scaled
- ▶ Convolution only applies for LTI systems

Convolution

- ▶ This operation = the **convolution** of two signals $x[n]$ and $h[n]$

$$x[n] * h[n] = \sum_{k=-\infty}^{\infty} x[k]h[n-k]$$

- ▶ The response of a LTI system to an input signal $x[n]$ is **the convolution of $x[n]$ with the system's impulse response $h[n]$**

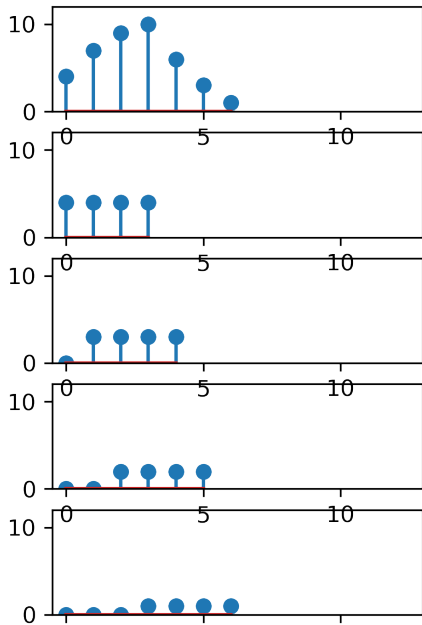
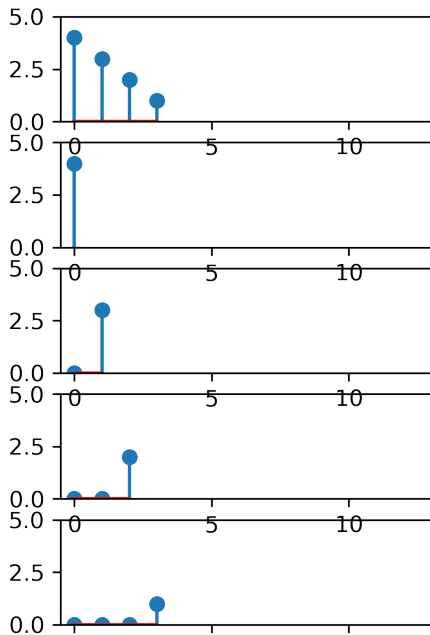
$$y[n] = x[n] * h[n]$$

Convolution

- ▶ Convolution is commutative: $x[n] * h[n] = h[n] * x[n]$
 - ▶ in equation it doesn't matter which signal has $[k]$ and which with $[n - k]$

$$x[n] * h[n] = \sum_{k=-\infty}^{\infty} x[k]h[n - k] = \sum_{k=-\infty}^{\infty} x[n - k]h[k]$$

Example



Interpretation of the convolution equation

The convolution equation can be interpreted in three ways:

1. The output signal $y[n]$ = a sum of a lot of impulse responses $h[n]$, each one delayed by k (hence $[n - k]$) and scaled by $x[k]$
 - ▶ one for each sample in the input signal
 - ▶ explain at blackboard

$$x[n] * h[n] = \sum_{k=-\infty}^{\infty} x[k] h[n - k]$$

Interpretation of the convolution equation

2. Each output sample $y[n]$ = a **weighted sum** of the input samples around it

$$y[n] = \dots + h[2] \cdot x[n-2] + h[1] \cdot x[n-1] + h[0] \cdot x[n] + h[-1] \cdot x[n+1] + \dots$$

- If $h[n]$ has finite length (e.g. non-zero only between $h[-2] \dots h[2]$), then there are only a few terms in the sum
- Example at blackboard

$$x[n] * h[n] = \mathbf{y[n]} = \sum_{k=-\infty}^{\infty} h[k] x[n-k]$$

Interpretation of the convolution equation

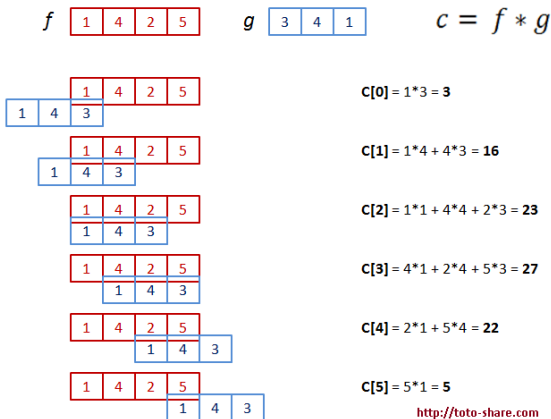


Figure 1: Convolution as weighted sum

► image from <http://www.stokastik.in>

Interpretation of the convolution equation

- ▶ Watch the following:

<https://www.youtube.com/watch?v=ulKbLD6BRJA>

Example

The impulse response can be read directly from the system equation (for non-recursive systems):

- ▶ Suppose we have the system:

$$y[n] = 3x[n+1] + 5x[n] - 2x[n-1] + 4x[n-2]$$

- ▶ What is the impulse response of the system?
- ▶ Answer: $h[n] = \{...0, 3, 5, -2, 4, 0, ...\}$

Convolution as matrix multiplication

- 3. Linear convolution can be written as multiplication with a **Toeplitz** matrix. Under periodic boundary conditions (circular convolution), the matrix becomes **circulant**.
- in this example, assuming $h[n]$ is non-zero only from $h[-1]$ to $h[3]$

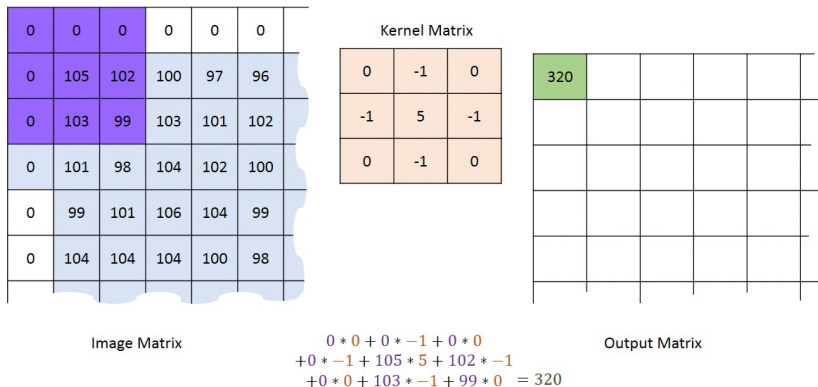
$$\begin{bmatrix} \vdots \\ y_n \\ y_{n+1} \\ y_{n+2} \\ y_{n+3} \\ \vdots \end{bmatrix} = \begin{bmatrix} \dots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \dots \\ \dots & 0 & h_3 & h_2 & h_1 & h_0 & h_{-1} & 0 & 0 & \dots \\ \dots & 0 & 0 & h_3 & h_2 & h_1 & h_0 & h_{-1} & 0 & \dots \\ \dots & 0 & 0 & 0 & h_3 & h_2 & h_1 & h_0 & h_{-1} & \dots \\ \dots & 0 & 0 & 0 & 0 & h_3 & h_2 & h_1 & h_0 & \dots \\ \dots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \dots \end{bmatrix} \cdot \begin{bmatrix} \vdots \\ x_{n-4} \\ x_{n-3} \\ x_{n-2} \\ x_{n-1} \\ x_n \\ x_{n+1} \\ x_{n+2} \\ x_{n+3} \\ \vdots \end{bmatrix}$$

2D convolution

- ▶ Convolution can be applied in 2D (for images)
- ▶ The input signal = an image $I[x, y]$
- ▶ The impulse response (the **kernel**) = a matrix $H[x, y]$
- ▶ The convolution result:

$$Y[x, y] = I * H = \sum_{i=-\infty}^{\infty} \sum_{j=-\infty}^{\infty} I[x - i, y - j] \cdot H[i, j]$$

2D convolution



**Convolution with horizontal and
vertical strides = 1**

Figure 2: 2D Convolution as weighted sum

2D Convolution

- ▶ Watch this:

http://machinelearningguru.com/computer_vision/basics/convolution/con

2D Convolution

- ▶ Simple image effects with 2D convolutions:
 - ▶ the “kernel” = the impulse response $H[x, y]$
- ▶ See here: [https://en.wikipedia.org/wiki/Kernel_\(image_processing\)](https://en.wikipedia.org/wiki/Kernel_(image_processing))
- ▶ What are their 1D counterparts?

Properties of convolution

Basic properties of convolution

- ▶ Convolution is **commutative** (the order of the signals doesn't matter):

$$x[n] * h[n] = \sum_{k=-\infty}^{\infty} x[k]h[n-k] = h[n] * x[n] = \sum_{k=-\infty}^{\infty} h[k]x[n-k]$$

- ▶ Proof: make variable change $(n-k) \rightarrow l$, change all in equation
- ▶ Convolution is **associative**:

$$(a[n] * b[n]) * c[n] = a[n] * (b[n] * c[n])$$

- ▶ (No proof)

Properties of convolution

- ▶ The unit impulse is **neutral element** for convolution:

$$a[n] * \delta[n] = \delta[n] * a[n] = a[n]$$

.

- ▶ Proof: equation

- ▶ Convolution is a **linear operation** (or **distributive**):

$$(\alpha \cdot a[n] + \beta \cdot b[n]) * c[n] = \alpha (a[n] * c[n]) + \beta (b[n] * c[n])$$

- ▶ Proof: by linearity of the corresponding system

Properties of LTI systems expressed with $h[n]$

1. Identity system

- ▶ A system with $h[n] = \delta[n]$ produces a response equal to the input, $y[n] = x[n], \forall x[n]$.
- ▶ Proof: $\delta[n]$ is neutral element for convolution.

Properties of LTI systems expressed with $h[n]$

2. Series connection is commutative

- ▶ LTI systems connected in series can be interchanged in any order
- ▶ Proof: by commutativity of convolution.
- ▶ LTI systems connected in series are equivalent to a single system with

$$h_{equiv}[n] = h_1[n] * h_2[n] * \dots * h_N[n]$$

Properties of LTI systems expressed with $h[n]$

3. Parallel connection means sum

LTI systems connected in parallel are equivalent to a single system with

$$h_{equiv}[n] = h_1[n] + h_2[n] + \dots + h_N[n]$$

Properties of LTI systems expressed with $h[n]$

4. Response of LTI systems to unit step

- ▶ If the input signal is $u[n]$, the response of the system is

$$s[n] = u[n] * h[n] = h[n] * u[n] = \sum_{k=-\infty}^{\infty} h[k]u[n-k] = \sum_{k=-\infty}^n h[k]$$

Properties of LTI systems expressed with $h[n]$

► Proof:

- The signal $\sum_{k=-\infty}^n h[k]$ is a *discrete-time integration* of $h[n]$
- The unit step $u[n]$ itself is the discrete-time integral of the unit impulse:

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

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

- Therefore the system response to the integral of the impulse = the integral of the system response to the impulse
- The interchanging of the integration with the system is due to the linearity of the system and is valid for all signals:

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

Relation between LTI system properties and $h[n]$

Relation between LTI system properties and $h[n]$

- ▶ For an LTI system, if we know $h[n]$, we know **everything** about the system
- ▶ Therefore, the properties (causal, memory, stability) must be reflected somehow in $h[n]$
 - ▶ Not linearity and time-invariance, they must be true, otherwise we wouldn't talk about $h[n]$

1. Causal LTI systems and their $h[n]$

If a LTI system is causal, then

$$h[n] = 0, \forall n < 0$$

► Proof:

- $y[n] = \sum_{k=-\infty}^{\infty} x[k]h[n-k]$,
- $y[n]$ does not depend on $x[n+1], x[n+2], \dots$
- it means that these terms are multiplied with 0
- the value $x[n+1]$ is multiplied with $h[n-(n+1)] = h[-1]$, $x[n+2]$ is multiplied with $h[n-(n+2)] = h[-2]$, and so on
- Therefore:

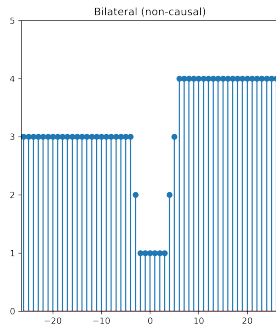
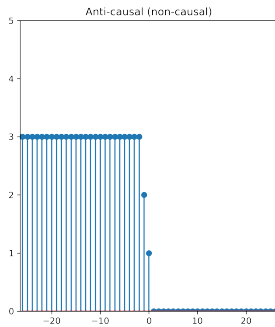
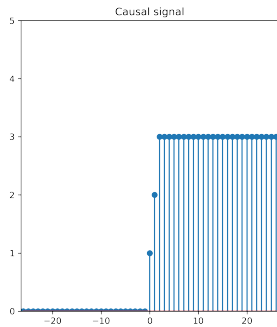
$$h[n] = 0, \forall n < 0$$

.

Causal signals and causal systems

- ▶ A **signal** which is 0 for $n < 0$ is called a **causal** signal
- ▶ Otherwise the signal is **non-causal**
- ▶ We can say that a **system** is causal if and only if it has a causal **impulse response**
- ▶ Further definitions:
 - ▶ a signal which is 0 for $n > 0$ is called an **anti-causal** signal
 - ▶ a signal which has non-zero values both for some $n > 0$ and for some $n < 0$ (and thus is neither causal nor non-causal) is called **bilateral**

Example



2. Stable systems and their $h[n]$

- ▶ Considering a bounded input signal, $|x[n]| \leq A$, the absolute value of the output is:

$$\begin{aligned}|y[n]| &= \left| \sum_{k=-\infty}^{\infty} x[k]h[n-k] \right| \\ &\leq \sum_{k=-\infty}^{\infty} |x[k]h[n-k]| \\ &= \sum_{k=-\infty}^{\infty} |x[k]| |h[n-k]| \\ &\leq A \sum_{k=-\infty}^{\infty} |h[n-k]| \end{aligned}$$

- ▶ Therefore a **LTI system is stable if**

$$\sum_{k=-\infty}^{\infty} |h[k]| < \infty$$

3. Memoryless systems and their $h[n]$ (Exercise)

Exercises:

- ▶ What can we say about the impulse response $h[n]$ of a memoryless system?
- ▶ What about a system with finite memory M ?

Hint/Answer:

- ▶ For an LTI memoryless system, $y[n] = c x[n]$, so $h[n] = c \delta[n]$.
- ▶ For an LTI system with finite memory M (causal), $h[n]$ has finite support (FIR), typically nonzero only for $n \in \{0, 1, \dots, M\}$.

FIR and IIR systems

- ▶ The **support** of a discrete signal = the smallest interval of n such that the signal is 0 everywhere outside the interval.
- ▶ Examples: at whiteboard
- ▶ Depending on the support of the impulse response, discrete LTI systems can be **FIR** or **IIR** systems.

FIR systems

- ▶ A **Finite Impulse Response (FIR)** system has an impulse response with **finite support**
 - ▶ i.e. the impulse response is 0 outside a certain interval.
 - ▶ i.e. $h[n]$ is zero beyond some element $h[M]$
- ▶ The system equation for a FIR system:

$$y[n] = \sum_{k=0}^M h[k]x[n-k] = h[0] \cdot x[n] + h[1] \cdot x[n-1] + \dots h[M] \cdot x[n-M]$$

- ▶ is non-recursive (depends only on x)
- ▶ goes only up to some term $h[M]x[n-M]$
- ▶ for causal system, starts from $h[0]x[n]$; for non-causal, can start from $h[-k]x[n+k]$

- ▶ For a causal FIR system, the output is a **linear combination** of **the last $M+1$ input samples**
- ▶ For non-causal FIR system, some future input samples enter the combination

IIR systems

- ▶ An **I**nfinite **I**mpulse **R**esponse (**IIR**) system has an impulse response with **infinite support**
 - ▶ i.e. the impulse response never becomes completely 0 forever.
- ▶ The output $y[n]$ potentially depends on all the preceding input samples
 - ▶ from the convolution equation:

$$\begin{aligned}y[n] &= \sum_{k=0}^{\infty} h[k]x[n-k] \\&= h[0] \cdot x[n] + h[1] \cdot x[n-1] + \dots h[M] \cdot x[n-M] + \dots \text{goes on} + \dots\end{aligned}$$

- ▶ An IIR system has infinite memory

IIR systems

- ▶ IIR systems must have **recursive** equations:
 - ▶ they depend on previous outputs $y[n-1]$ up to $y[n-N]$
 - ▶ they also depend on input, going back up to $x[n-k]$
- ▶ General equation of an IIR system:

$$\begin{aligned}y[n] &= -a_1y[n-1] - a_2y[n-2] - \dots - a_Ny[n-N] + \\&\quad + b_0x[n] + b_1x[n-1] + \dots + b_Mx[n-M] \\&= -\sum_{k=1}^N a_k y[n-k] + \sum_{k=0}^M b_k x[n-k]\end{aligned}$$

- ▶ the impulse response cannot be read explicitly from the equation
- ▶ IIR equations are more general than FIR

General equation of an LTI system

Recap:

- ▶ The general equation of an LTI system is:

$$\begin{aligned}y[n] &= -a_1y[n-1] - a_2y[n-2] - \dots - a_Ny[n-N] + \\&\quad + b_0x[n] + b_1x[n-1] + \dots + b_Mx[n-M] \\&= -\sum_{k=1}^N a_k y[n-k] + \sum_{k=0}^M b_k x[n-k]\end{aligned}$$

- ▶ If all $a_i = 0$, it is a FIR system, no $y[n-k]$ term
 - ▶ in this case the coefficients $b_k = h[k]$ (impulse response)
- ▶ If some $a_i \neq 0$, it is an IIR system
 - ▶ impulse response $h[n]$ is infinitely long, is more complicated to find
- ▶ Note: if system is non-causal, can start from $x[n+k]$

Initial conditions for recursive systems

- ▶ Recursive systems need **initial conditions** (starting values)
 - ▶ since they rely on previous outputs
- ▶ If initial conditions are all 0, the system is **relaxed**
 - ▶ the output depends only on the input signal
- ▶ If initial conditions are not zero, the output depends on the input signal **and** the initial conditions

Initial conditions for recursive systems

- ▶ The effect of the input signal and the effect of initial conditions are **independent**
 - ▶ the system behaves **linear** with respect to them
 - ▶ total output = output due to input + output due to initial conditions

Input	Init.Cond.	O utput
$x[n]$	0	$y[n] = y_{zs}[n]$
0	non-zero	$y[n] = y_{zi}[n]$
$x[n]$	non-zero	$y[n] = y_{zs}[n] + y_{zi}[n]$

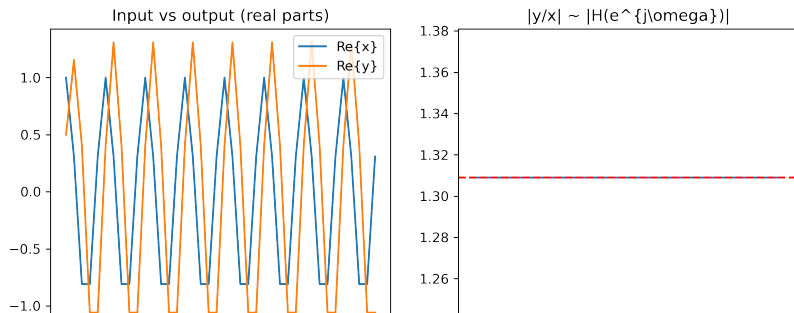
Complex exponentials as LTI eigenfunctions

Complex exponentials as LTI eigenfunctions

- For an LTI system with impulse response $h[n]$, a complex exponential input $x[n] = z^n$ produces an output

$$y[n] = (h * x)[n] = \sum_{k=-\infty}^{\infty} h[k] z^{n-k} = \underbrace{\left(\sum_{k=-\infty}^{\infty} h[k] z^{-k} \right)}_{H(z)} z^n.$$

Thus, $e^{j\omega n}$ is an eigenfunction with eigenvalue $H(e^{j\omega})$ (the frequency response).



Moving-average FIR example

Moving-average FIR example

- ▶ A 3-point moving average has $h[n] = \frac{1}{3}\{1, 1, 1\}$ and smooths noise.

