

# Discrete-time Signal Processing

MsCV Vibot - UE4 Digital Signal Processing

Olivier Morel

Le2i Vision Robotics Team - ERL CNRS

December 11, 2017

# Preamble

## ■ Main references:



*Digital Signal Processing, Principles, Algorithms, and Applications*,  
John G. Proakis, Dimitris G. Manolakis.



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- 8x2H Main course mix with tutorials
- Assessment : closed book

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⇒ 1 sheet of A4 paper allowed!

- double-sided

- manuscript

- no hard copy

- 1 Introduction**
- 2 Discrete-Time signals
- 3 Discrete-Time systems
- 4 The Z-Transform (ZT)
- 5 Fourier Transform of DT signals
- 6 Analysis of LTI systems in the z-Domain
- 7 Discrete Fourier Transform (DFT)

# Introduction

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  - **Finance**: time variations of a stock value or a market index.
- By extension, any series of measurements of a physical quantity can be considered a signal (temperature measurements for instance)

# Types of signals and representations

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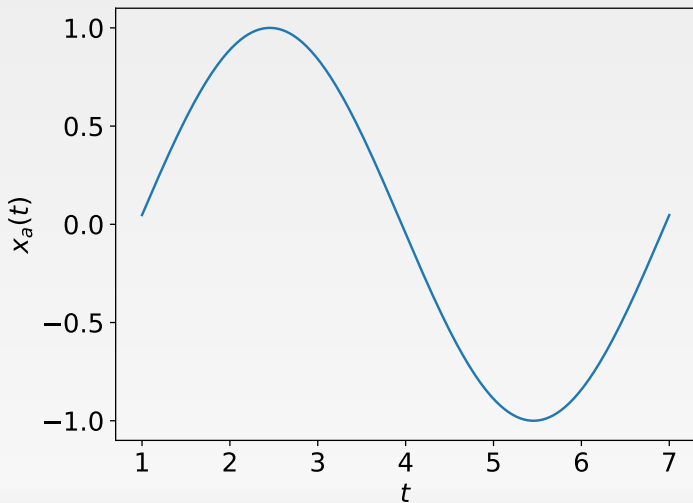
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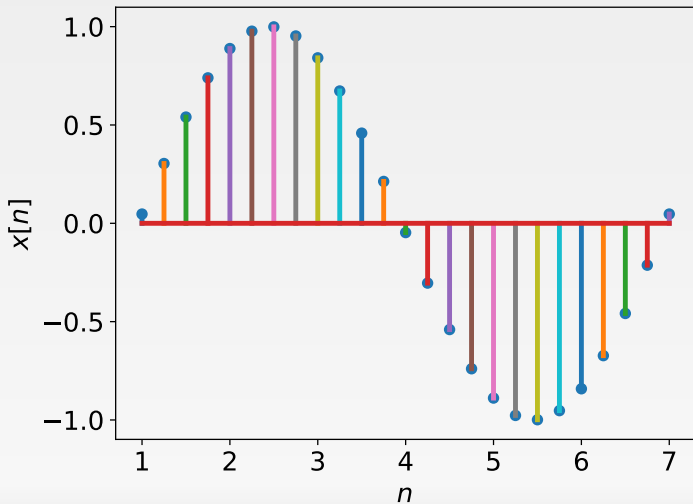
## Multi-dimensional signal

$$x(t_1, \dots, t_N)$$

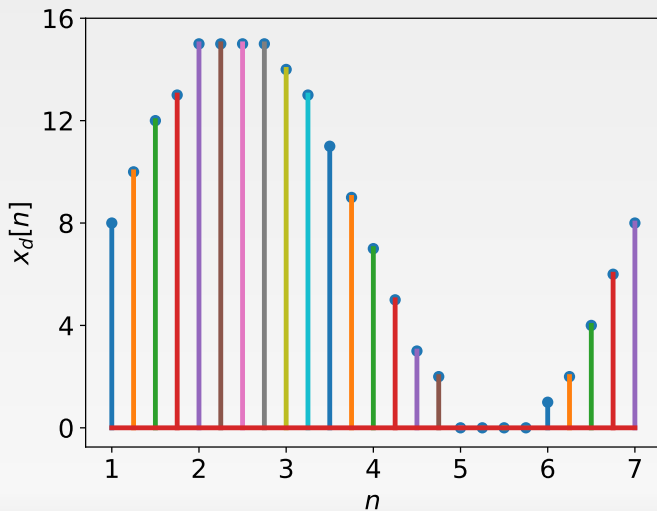
## Example: analog signal



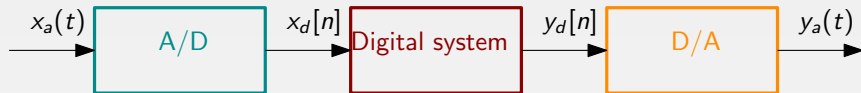
# Example: discrete signal



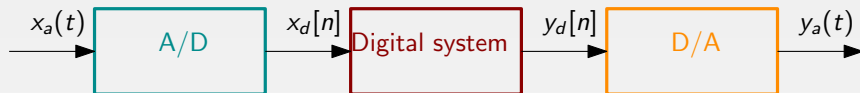
# Example: digital signal



# Generic structure of a DSP

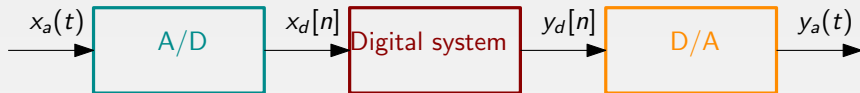


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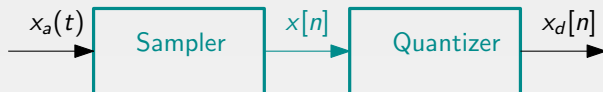
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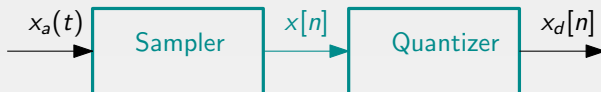
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3. The D/A (digital-to-analog) converter transforms the digital output into an analog signal  $y_a(t)$



# A/D converter



# A/D converter



- **Sampler:** analog input is transformed into a DT signal  $x[n] = x_a(nT_s)$  where  $T_s$  is the sampling period.

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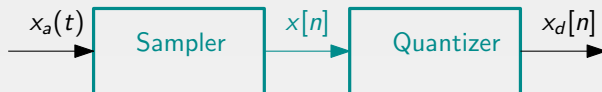
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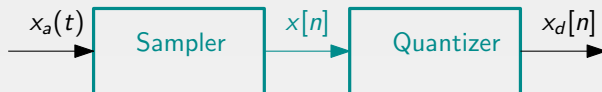
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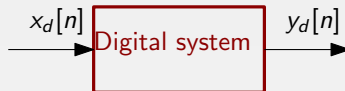
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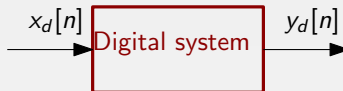
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  - level resolution: 16 bits per sample (some systems use 24 bits)



# Digital system

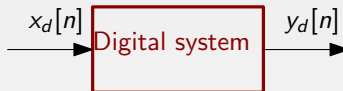


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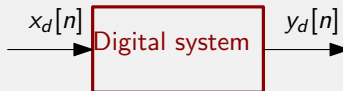
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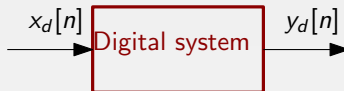
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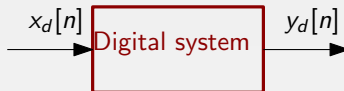
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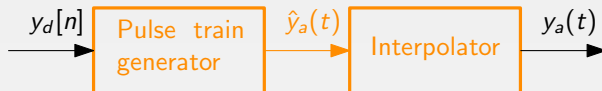
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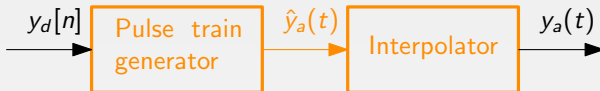


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  - **Off-line system**: requires the use of external data storage units

# D/A converter



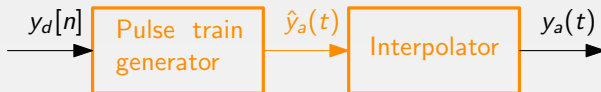
# D/A converter



- **Pulse train generator:** the digital signal  $y_d[n]$  is transformed into a sequence of scaled, analog pulses

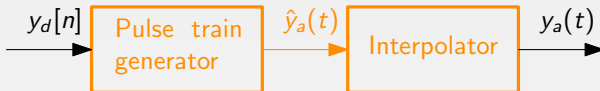


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- One device can generally take care of both steps.

# Pros and cons of DSP

## Advantages

- Robustness (signal levels can be regenerated)
- Storage capability (can interfaced to low-cost devices for storage)
- Flexibility (software programmable)
- Structure (easy interconnection of DSP blocks)

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## Disadvantages

- Cost/complexity added by A/D and D/A conversion
- Input signal bandwidth is technology limited
- Quantization effects

- 1 Introduction
- 2 Discrete-Time signals**
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## Definition

sequence of real or complex numbers, that is, a mapping from the set of integers  $\mathbb{Z}$  into  $\mathbb{R}$  or  $\mathbb{C}$ , as in:

$$n \in \mathbb{Z} \rightarrow x[n] \in \mathbb{R} \text{ or } \mathbb{C}$$

- $n$  is called the discrete-time index
- $x[n]$ , the  $n$ th number in the sequence, is called a sample

# Description

- Sequence notation:

$$x = \{\dots, 0, \underline{0}, 1, 4, 1, 0, 0, \dots\},$$

where underline indicates origin of time:  $n = 0$

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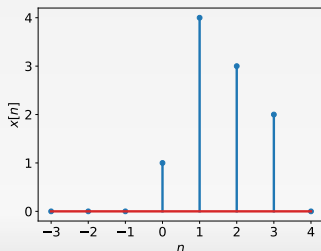
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- Graphical:



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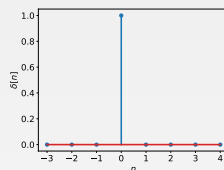
- Recursive approach:

$$x[n] = \begin{cases} 0 & n < 0, \\ 1 & n = 0, \\ \frac{1}{2}x[n-1] & n > 0. \end{cases}$$

# Basic Discrete-time signals

- Unit pulse:

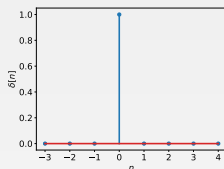
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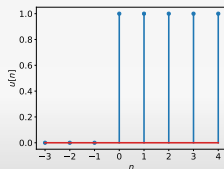
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## ■ Unit step:

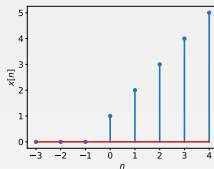
$$u[n] = \begin{cases} 1 & n \geq 0, \\ 0 & n < 0. \end{cases}$$



# Basic Discrete-time signals

- Ramp function:

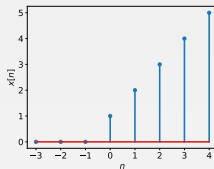
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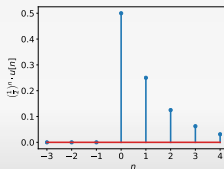
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- Exponential sequence:

$$x[n] = a^n \cdot u[n].$$



# Uniform sampling

- DT signals are commonly generated via uniform (or periodic) sampling of an analog signal  $x_a(t)$ :

$$x[n] = x_a(nT_s), \quad n \in \mathbb{Z},$$

where  $T_s > 0$  : sampling period.



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- Example of a sine wave with frequency  $F$  defined by  $x_a(t) = \sin 2\pi Ft$

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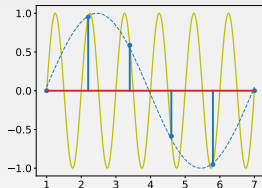
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- $F_s = 1/T_s$ : sampling frequency,
- $\omega$  : normalized radian frequency of the DT signal.

# Nyquist rate



**Nyquist rate:  $2 \cdot F$**

The sampling frequency  $F_s$  must satisfy

$$F_s \gg 2 \cdot F$$

# Basic operations on signal

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## Property

$\mathcal{S}$  equipped with addition and scaling is a vector space.



# Classes of signals

## Energy signals

all  $x \in \mathcal{S}$  with finite energy:

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## Power signals

all  $x \in \mathcal{S}$  with finite power:

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

# Classes of signals

## Bounded signals

all  $x \in \mathcal{S}$  that can be bounded:

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## Absolutely summable

all  $x \in \mathcal{S}$  such that:

$$\sum_{n=-\infty}^{\infty} |x[n]| < \infty$$

# Discrete convolution

## Discrete convolution of $x$ and $y$

$$(x * y)[n] = x[n] * y[n] \triangleq \sum_{k=-\infty}^{\infty} x[k] \cdot y[n - k]$$

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## Properties

- commutative law:  $x * y = y * x$
- associative law:  $(x * y) * z = x * (y * z)$
- convolution by unit pulse:  $x * \delta = x$

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- **Autocorrelation**: performed on one signal
  - indicates how the signal energy (power) is distributed within the signal
  - applications of signal autocorrelation are in radar, sonar, satellite, and wireless communications systems

# Crosscorrelation

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

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## Properties

- Even function:

$$R_{xx}[n] = R_{xx}[-n]$$

- Energy:

$$R_{xx}[0] = \sum_{k=-\infty}^{\infty} |x[k]|^2 = \mathcal{E}_x$$

$$\forall n \in \mathbb{Z}, R_{xx}[n] < R_{xx}[0]$$

# Correlation coefficient

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## Properties

- Similarity measurement of two signals
- $-1 \leq c_{xy} \leq 1$
- Geometrically represents angle between euclidean vectors  $x$  and  $y$

$$C_{xy} = \frac{x \cdot y}{\sqrt{|x|^2 |y|^2}} = \frac{x \cdot y}{|x| |y|} \triangleq \cos(x, y)$$

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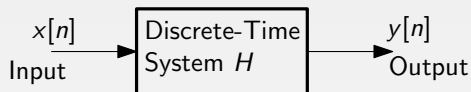
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- An also be defined in terms of paramter  $n$ :

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- 1 Introduction
- 2 Discrete-Time signals
- 3 Discrete-Time systems**
- 4 The Z-Transform (ZT)
- 5 Fourier Transform of DT signals
- 6 Analysis of LTI systems in the z-Domain
- 7 Discrete Fourier Transform (DFT)

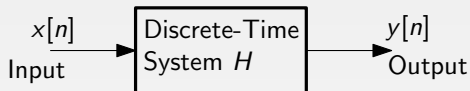
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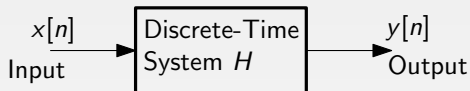


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- Notations:

$$y[n] = H(x[n]) \triangleq (Hx)[n]$$

## 3 Discrete-Time systems

- Basic systems
- Linear Time-Invariant (LTI) systems

# Basic systems

## Time reversal

### Time reversal

$$y[n] = (Rx)[n] \triangleq x[-n]$$

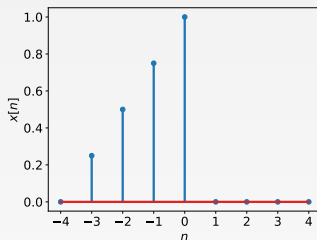
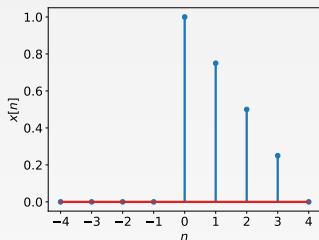
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- Mirror image about origin:



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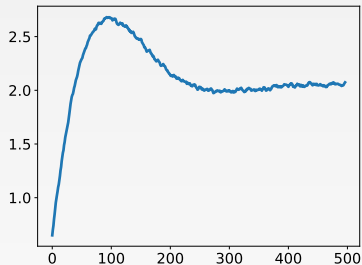
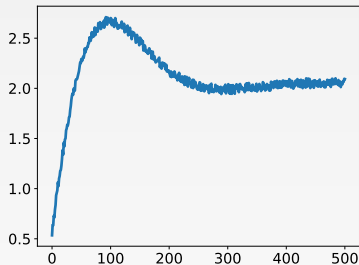
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- Dynamic systems require memory:

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

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## Anti-causal

$y[n]$  only depends on values  $x[k]$  for  $k > n$ .

# Systems properties: linear or not ?

## Linearity

$$\forall (\alpha, \beta) \in \mathbb{C}^2, \forall (x, y) \in \mathcal{S}^2, H(\alpha x + \beta y) = \alpha H(x) + \beta H(y)$$

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- A system is stable (Bounded Input Bounded Output) if every bounded input produces a bounded output.

## 3 Discrete-Time systems

- Basic systems
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# Linear Time-Invariant (LTI) systems

- DT systems that are both **Linear** and **Time-Invariant** play a central role in digital signal processing:
  - Many physical systems are either LTI or approximately so
  - Many efficient tools are available for the analysis and design of LTI systems



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## Fundamental property

Let  $H$  a LTI system and  $y = Hx$

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

with  $h \triangleq H\delta$  known as impulse response of  $H$ .

# Proof of the fundamental property

First we have:

$$y[n] = (Hx)[n] = H(x[n]).$$

And for any DT signal, we can write:

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Invoking Time-Invariant property:

$$y[n] = \sum_{k=-\infty}^{\infty} x[k]HD_k(\delta[n]) = \sum_{k=-\infty}^{\infty} x[k]D_kH(\delta[n]) = \sum_{k=-\infty}^{\infty} x[k]D_kh[n]$$

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- Example with  $h = \{\dots, 0, 1, \underline{0}, -1, 0, \dots\}$  and  $x = u$ :

$x[n]$	0	0	0	<b>0</b>	1	<b>1</b>	1	1	1
mask			$\rightarrow$	<b>-1</b>	0	<b>1</b>	$\rightarrow$		
$y[n]$	0	0	0	1	<b>1</b>	0	0	0	0

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$\Rightarrow$  Clearly,  $y[n]$  only depends on values  $x[m]$  for  $m \leq n$  if and only if  $h[k] = 0$  for  $k < 0$

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Thus the system is **stable** provided  $|\alpha| < 1$



## FIR system

An LTI system has a **Finite Impulse Response (FIR)** if we can find integers  $N_1 \leq N_2$  such that:

$$h[n] = 0 \text{ when } n < N_1 \text{ or } n > N_2$$

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- The impulse response is often called a **convolution mask**.

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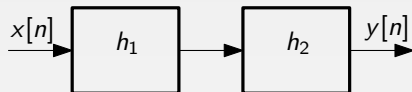
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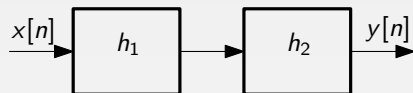
$\Rightarrow$  the system is **IIR** (cannot find any  $N_2$ )



# Interconnection of LTI systems: cascade

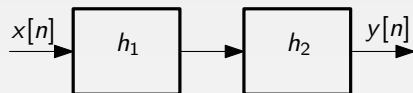


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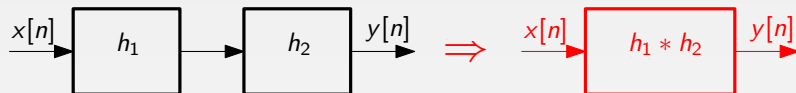
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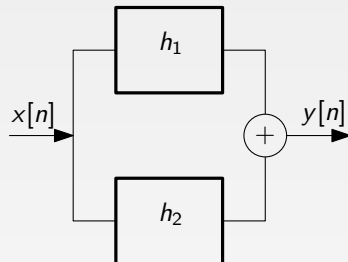
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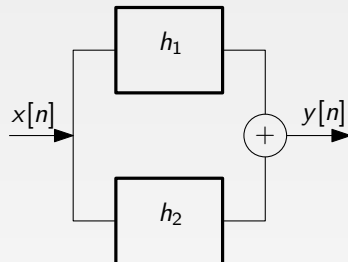


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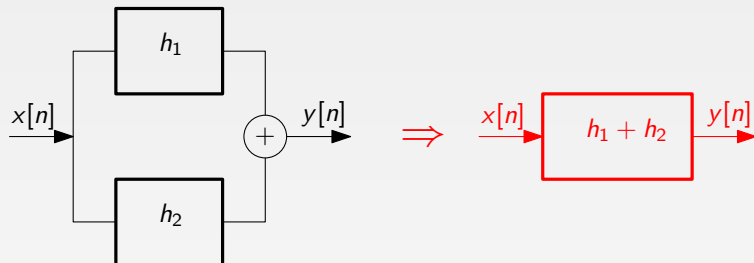


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- 1 Introduction
- 2 Discrete-Time signals
- 3 Discrete-Time systems
- 4 The Z-Transform (ZT)**
- 5 Fourier Transform of DT signals
- 6 Analysis of LTI systems in the z-Domain
- 7 Discrete Fourier Transform (DFT)



## 4 The Z-Transform (ZT)

- Definition
- Study of the ROC
- Properties of the ZT
- Rational ZTs
- Inverse ZT

# Introduction

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- **Both  $X(z)$  and the ROC are needed** when specifying a ZT.

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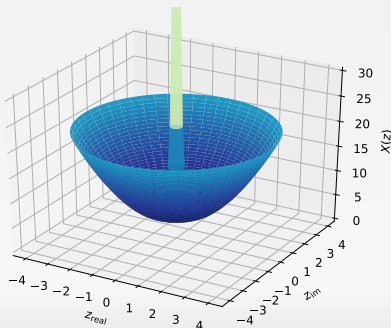
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# Theorem

## Radius of convergence

To any power series  $\sum_{n=0}^{\infty} c_n w^n$ , we can associate a **radius of convergence**

$$R_w = \lim_{n \rightarrow \infty} \left| \frac{c_n}{c_{n+1}} \right|,$$

such that:

- if  $|w| < R_w \Rightarrow$  the series converges absolutely
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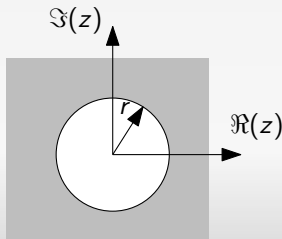
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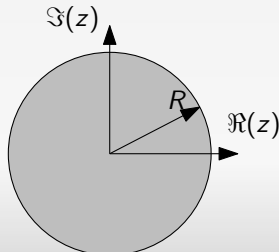
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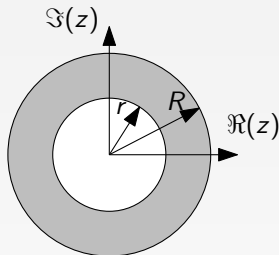
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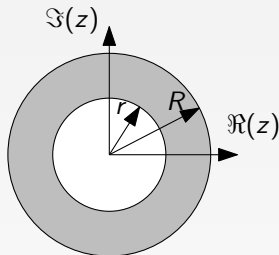


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The two series will converge iff ROC:  $\frac{1}{2} < |z| < 2$

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Since  $\{|z| > 2\} \cap \{|z| < \frac{1}{2}\} = \emptyset$ , the ROC is empty and the ZT does not exist.

## 4 The Z-Transform (ZT)

- Definition
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# Introductory remarks

- Notations for ZT pairs:

$$x[n] \xleftrightarrow{z} X(z), z \in \mathcal{R}_x$$

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- In some cases, the true ROC may be larger than the one indicated

# Basic symmetries

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Assuming that the ROC  $\mathcal{R}_x$  was defined by:  $r < |z| < R$ , then the ROC of  $X_f(z)$  is:  $1/R < |z| < 1/r$

# Linearity and Time shift

## Linearity

$$\forall (a, b) \in \mathbb{C}^2, ax[n] + by[n] \xleftrightarrow{z} aX(z) + bY(z), z \in \mathcal{R}_x \cap \mathcal{R}_y$$

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$$X(z) = \frac{1 - z^{-1} \cos \omega_0}{1 - 2z^{-1} \cos \omega_0 + z^{-2}}, \text{ ROC : } |z| > 1$$

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# Initial value

## Initial value (causal case)

For  $x[n]$  causal (i.e.  $x[n] = 0$  for  $n < 0$ ), we have:

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For  $x[n]$  anti-causal (i.e.  $x[n] = 0$  for  $n > 0$ ), we have:

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## 4 The Z-Transform (ZT)

- Definition
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# Rational function

## Definition

$X(z)$  is a rational function in  $z$  (or  $z^{-1}$ ) if:

$$X(z) = \frac{N(z)}{D(z)}$$

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- Rational ZT plays a central role in DSP
- Essential for the realization of practical IIR filters
- Two important issues related to rational ZT are investigated:
  - Pole-Zero (PZ) characterization
  - Inversion via partial fraction expansion

# Poles and zeros

## Pole

$X(z)$  has a **pole** of order  $L$  at  $z = p_0$  if:

$$X(z) = \frac{\psi(z)}{(z - p_0)^L}, \quad 0 < |\psi(p_0)| < \infty$$

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- The order  $L$  is sometimes referred as the **multiplicity** of the pole/zero.



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## Property

number of poles = number of zeros, if poles and zeros at 0 and  $\infty$  are included.

# Poles and Zeros of a rational: examples

## ■ Example 1:

$$X(z) = \frac{z^{-1}}{1 - 2z^{-1} + z^{-2}} = \frac{z}{z^2 - 2z + 1} = \frac{z}{(z - 1)^2}$$

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 $p_2 = -3, L = 1$



# Poles and Zeros of a rational: examples

## ■ Example 1:

$$X(z) = \frac{z^{-1}}{1 - 2z^{-1} + z^{-2}} = \frac{z}{z^2 - 2z + 1} = \frac{z}{(z - 1)^2}$$

Corresponding poles and zeros:

**poles**  $p_1 = 1, L = 2$

**zeros**  $z_1 = 0, L = 1$   
 $z_2 = \infty, L = 1$

## ■ Example 2:

$$X(z) = \frac{1 - z^{-4}}{1 + 3z^{-1}} = \frac{z^4 - 1}{z^3(z + 3)}$$

Corresponding poles and zeros:

**poles**  $p_1 = 0, L = 3$   
 $p_2 = -3, L = 1$

**zeros**  $z_{k \in [0,3]} = e^{jk\pi/2}, L = 1$

# Pole-zero and rational function link

## Property

For rational functions  $X(z) = N(z)/D(z)$ , knowledge of the poles and zeros (along with their order) completely specify  $X(z)$ , up to a scaling factor  $G \in \mathbb{C}$ .

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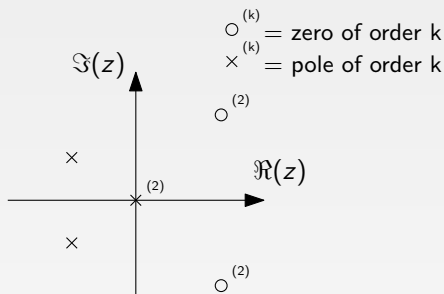
### ■ Example:

**poles**  $p_1 = 2, L = 1$

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$$X(z) = G \frac{z - 1}{z - 2} = G \frac{1 - z^{-1}}{1 - 2z^{-1}}$$

# Pole-zero (PZ) diagram



- The presence of poles or zeros at  $\infty$  should be mentioned on the diagram
- It is useful to indicate ROC on the PZ-diagram

# ROC and PZ diagram

- Consider  $x[n] = a^n u[n]$ , where  $a > 0$ :

$$X(z) = \frac{1}{1 - az^{-1}} = \frac{z}{z - a}, \text{ ROC: } |z| > a$$

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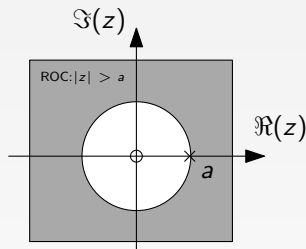
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  - accordingly, several possible DT signals  $x[n]$



## 4 The Z-Transform (ZT)

- Definition
- Study of the ROC
- Properties of the ZT
- Rational ZTs
- Inverse ZT

# Introduction

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  - Power series expansion
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- **Partial fraction** is the most useful technique in the context of **rational ZTs**

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## Inverse Z-Transform

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- Cauchy integral theorem:

$$\frac{1}{2\pi j} \oint_C z^{k-1-n} dz = \begin{cases} 1 & \text{if } k = n \\ 0 & \text{if } k \neq n \end{cases}$$

# Inversion via Partial Fraction Expansion

- Let be a rational ZT defined according to:
  - $X(z) = \frac{N(z)}{D(z)}$
  - $N(z)$  and  $D(z)$  are polynomials in  $z^{-1}$
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- Under these conditions,  $X(z)$  may be expressed as:

$$X(z) = \sum_{k=1}^K \sum_{l=1}^{L_K} \frac{A_{kl}}{(1 - p_k z^{-1})^l}$$

- $p_1, \dots, p_K$  are the distinct poles of  $X(z)$
- $L_1, \dots, L_K$  are the corresponding orders

## Expression of the constants $A_{kl}$

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Given  $X(z)$  as above with ROC:  $r < |z| < R$ .

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- $\frac{1}{1-az^{-1}} \xrightarrow{\mathcal{Z}^{-1}} a^n u[n]$  since  $|z| > |a|$  (causal)
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- Finally:

$$x[n] = \frac{a^{n+1}}{a-b} u[n] - \frac{b^{n+1}}{b-a} u[-n-1]$$



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  - use of shift property

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- To decrease the  $z^{-1}$  power in  $N(z)$ ,  $D(z)$  and  $N(z)$  are expressed in increasing powers of  $z$  (e.g.  $D(z) = z^{-2} + 2z^{-1} + 1$ )

# Example

ZT of a causal signal

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$$X(z) \text{ rewrites } X(z) = 1 - \frac{z^{-1} + 8}{z^{-2} + 4z^{-1} + 3}$$

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$$X(z) = 1 - \frac{z^{-1} + 8}{z^{-2} + 4z^{-1} + 3}$$

- The denominator of the 2nd term has 2 roots, poles at  $z = -1/3$  and  $z = -1$ , hence:

$$X(z) = 1 - \frac{z^{-1} + 8}{3 \left(1 + \frac{1}{3}z^{-1}\right) (1 + z^{-1})}$$

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- The PFE gives:

$$X(z) = 1 - \frac{1}{3} \left( \frac{A_1}{1 + \frac{1}{3}z^{-1}} + \frac{A_2}{1 + z^{-1}} \right)$$

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  - causal sequence  $\Rightarrow$  ROC must extend outwards from the outermost pole  $\Rightarrow$  ROC is  $|z| > 1$
- The sequence  $x[n]$  is then given by:

$$x[n] = \delta[n] + \frac{5}{6} \left(-\frac{1}{3}\right)^n u[n] - \frac{7}{2} (-1)^n u[n]$$

# Use of shift property

- In some cases, a simple multiplication by  $z^k$  is sufficient to put  $X(z)$  into a suitable format, that is:

$$Y(z) = z^k X(z) = \frac{N(z)}{D(z)}$$

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- Finally, the shift property is applied to recover  $x[n]$ :

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

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$$X(z) = \frac{1 - z^{-128}}{1 - z^{-2}}, \quad |z| > 1$$

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- Therefore:

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## 5 Fourier Transform of DT signals

- Definition
- Convergence of the DTFT
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  - $|X(\omega)|$ : magnitude spectrum
  - $\angle X(\omega)$ : phase spectrum
- The Fourier Transform is a specific case of the ZT taking  $z = e^{j\omega}$  with  $|z| = 1 \in \text{ROC}$ .

# Fourier Transform of a sampled continuous signal I

- Let  $s_e(t)$  the sampled expression of the continuous signal  $s(t)$  with sampling period  $T_s$ :

$$s_e(t) = s(t) \cdot \sum_{k=-\infty}^{+\infty} \delta(t - kT_s)$$

- If we denote  $s[k] = s(kT_s)$ , we have:

$$s_e(t) = \sum_{k=-\infty}^{\infty} s[k] \delta(t - kT_s)$$

- The Fourier transform gives:

$$TF\{s_e(t)\} = \hat{s}_e(\omega) = s[k] \sum_{k=-\infty}^{\infty} s[k] TF\{\delta(t - kT_s)\}$$

# Fourier Transform of a sampled continuous signal II

- Applying the delay theorem:

$$\hat{s}_e(\omega) = s[k] \sum_{k=-\infty}^{\infty} s[k] e^{-jk\omega T_s} TF\{\delta(t)\} = s[k] \sum_{k=-\infty}^{\infty} s[k] e^{-jk\omega T_s}$$

- Finally:

$$\hat{s}_e(\omega) = S(\omega T_s)$$

with the Nyquist frequency equal to:

$$\omega_N = \frac{\pi}{T_s}$$



## Inverse DTFT

$$x[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(\omega) e^{j\omega n} d\omega, \quad n \in \mathbb{Z}$$

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$$x[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(\omega) e^{j\omega n} d\omega, \quad n \in \mathbb{Z}$$

Proof: Note that  $\int_{-\pi}^{\pi} e^{j\omega n} d\omega = 2\pi\delta[n]$ :

$$\begin{aligned} \int_{-\pi}^{\pi} X(\omega) e^{j\omega n} d\omega &= \int_{-\pi}^{\pi} \left( \sum_{k=-\infty}^{\infty} x[k] e^{-j\omega k} \right) e^{j\omega n} d\omega \\ &= \sum_{k=-\infty}^{\infty} x[k] \int_{-\pi}^{\pi} e^{j\omega(n-k)} d\omega \\ &= 2\pi \sum_{k=-\infty}^{\infty} x[k] \delta[n-k] \\ &= 2\pi x[n] \end{aligned}$$

## 5 Fourier Transform of DT signals

- Definition
- Convergence of the DTFT
- Properties

# Convergence of the DTFT

- For the DTFT to exist, the series  $\sum_{n=-\infty}^{\infty} x[n]e^{-j\omega n}$  must converge
- That is, the partial sum

$$X_M(\omega) = \sum_{n=-M}^M x[n]e^{-j\omega n}$$

must converge to a limit  $X(\omega)$  as  $M \rightarrow \infty$

- **Absolutely summable signals**

- $X_M(\omega)$  converges uniformly to  $X(\omega)$
- $X(\omega)$  is continuous

- **Energy signals**

- $X_M(\omega)$  does not necessarily converge
- $X(\omega)$  may be discontinuous at certain points

- **Power signals**

- Most power signals do not have a DTFT
- Exceptions including: Periodic signals, Unit step

## 5 Fourier Transform of DT signals

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- Differentiation

$$nx[n] \xleftrightarrow{\mathcal{F}} j \frac{dX(\omega)}{d\omega}$$

# Even and odd component definition

## DT signal

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

$$x_e[n] \triangleq \frac{1}{2} (x[n] + x^*[-n]) = x_e^*[-n]$$

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$$X(\omega) = X_e(\omega) + X_o(\omega)$$

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# Basic symmetries

## Real and imaginary parts decomposition

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## 6 Analysis of LTI systems in the z-Domain

- LTI systems described by LCCDE
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# Linear Constant Coefficient Difference Equations

## Definition

A DT system can be described by an LCCDE of order  $N$  if:

$$\sum_{k=0}^N a_k y[n-k] = \sum_{k=0}^M b_k x[n-k]$$

where  $a_0 \neq 0$  and  $a_N \neq 0$ .

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where  $a_0 \neq 0$  and  $a_N \neq 0$ .

- If we further assume initial **rest conditions**, i.e.:

$$\forall n < n_0, x[n] = 0 \Rightarrow \forall n < n_0, y[n] = 0$$

LCCDE corresponds to unique causal LTI system.

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LCCDEs lead to efficient recursive implementation:

- Recursive because computation of  $y[n]$  make use past output signal values ( $y[n-1]$ )
- Efficient: in the case of the accumulator it requires only 1 adder and 1 memory unit instead of an infinite number of adders and memory units.

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# One-sided Z-Transform

- The two-sided ZT requires that the corresponding signals be specified for entire time range  $-\infty < n < \infty$ 
  - Prevent evaluation of the output of non-relaxed systems
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## Characteristics

1. Does not contain information about the signal  $x[n]$  for negative values of time ( $n < 0$ )
2. It is **unique** only for **causal** signals
3. one-sided ZT of  $x[n]$  is identical to the two-sided ZT of  $x[n]u[n]$

# Properties

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## Case 1: Time Delay

If

$$x[n] \xleftrightarrow{z^+} X^+(z)$$

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$$\forall k > 0, x[n - k] \xleftrightarrow{z^+} z^{-k} \left( X^+(z) + \sum_{n=1}^k x[-n]z^n \right)$$

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## Case 2: Time advance

If

$$x[n] \xleftrightarrow{z^+} X^+(z)$$

then

$$\forall k > 0, x[n + k] \xleftrightarrow{z^+} z^k \left( X^+(z) - \sum_{n=0}^{k-1} x[n]z^{-n} \right)$$

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Determine the step response of the system with IC  $y[-1] = 1$ :

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1. By taking the one-sided ZT of both sides:

$$Y^+(z) = \alpha z^{-1} (Y^+(z) + y[-1]z) + X^+(z).$$

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Determine the step response of the system with IC  $y[-1] = 1$ :

$$y[n] = \alpha y[n-1] + x[n], \text{ with } -1 < \alpha < 1$$

1. By taking the one-sided ZT of both sides:

$$Y^+(z) = \alpha z^{-1} (Y^+(z) + y[-1]z) + X^+(z).$$

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5. Compute inverse ZT:

$$\begin{aligned} y[n] &= \alpha^{n+1} u[n] + \frac{1 - \alpha^{n+1}}{1 - \alpha} u[n] \\ &= \frac{1}{1 - \alpha} (1 - \alpha^{n+2}) u[n] \end{aligned}$$

# Final Value Theorem

If

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- The limit exists if the ROC of  $(z - 1)X^+(z)$  includes the unit circle
- Useful when the asymptotic behavior of a signal  $x[n]$  is desired knowing its ZT

## 6 Analysis of LTI systems in the z-Domain

- LTI systems described by LCCDE
- One-sided Z-Transform
- The system function
- Response of rational system Functions
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# The system function

LTI system  $\mathcal{H}$  (recall)

$$y[n] = x[n] * h[n] = \sum_{k=-\infty}^{\infty} x[k]h[n-k] \text{ with } h[n] = \mathcal{H}\{\delta[n]\}$$

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The system function of  $\mathcal{H}$ , denoted  $H(z)$  is the ZT of  $h[n]$ :

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where  $\mathcal{R}_H$  denotes the corresponding ROC.

- If  $H(z)$  and  $\mathcal{R}_H$  are known,  $h[n]$  can be recovered via inverse ZT
- if  $z = e^{j\omega} \in \mathcal{R}_H$  (the ROC contains the unit circle) then

$$H(e^{j\omega}) = \sum_{n=-\infty}^{\infty} h[n]e^{-j\omega n} \equiv H(\omega)$$

# Properties

Let  $\mathcal{H}$  be LTI system with system function  $H(z)$  and ROC  $\mathcal{R}_H$ .

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- LTI system  $\mathcal{H}$  is stable iff  $\mathcal{R}_H$  contains the unit circle:

$$\begin{aligned}\mathcal{H} \text{ stable} &\Leftrightarrow \sum_n |h[n]| < \infty \\ &\Leftrightarrow e^{j\omega} \in \mathcal{R}_H\end{aligned}$$

$$|H(z)| \leq \sum_n |h[n]z^{-n}|$$

evaluated on the unit circle:  $z = e^{j\omega}$ :

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# LCCDE system function

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Leading to a rational system function:

$$H(z) = \frac{Y(z)}{X(z)} = \frac{\sum_{k=0}^M b_k z^{-k}}{\sum_{k=0}^N a_k z^{-k}}$$



# Rational system with real coefficients

$$H(z) = \frac{\sum_{k=0}^M b_k z^{-k}}{\sum_{k=0}^N a_k z^{-k}}$$

- In many applications, coefficients  $a_k$  and  $b_k$  are real implying:

$$H^*(z) = H(z^*)$$

- Thus, if  $z_k$  is a zero of  $H(z)$  then:

$$H(z_k^*) = (H(z_k))^* = 0^* = 0$$

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- More generally, it can be shown that complex poles (or zeros) occur in complex conjugate pairs:
  - if  $p_k$  is a pole of order  $l$  of  $H(z)$ , so is  $p_k^*$
  - if  $z_k$  is a zero of order  $l$  of  $H(z)$ , so is  $z_k^*$

## **6** Analysis of LTI systems in the z-Domain

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# Response of rational system functions

- Let be  $H(z) = \frac{B(z)}{A(z)}$  the system function of a LCCDE system:
  - roots of  $A(z)$  are the poles of  $H(z)$
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- The partial fraction expansion of  $Y(z)$  yields

$$Y(z) = \sum_{k=1}^N \frac{A_k}{1 - p_k z^{-1}} + \sum_{k=1}^L \frac{Q_k}{1 - q_k z^{-1}}$$

if  $\forall k \forall m, p_k \neq q_m$  and there is no pole-zero cancellation.

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$$y[n] = \underbrace{\sum_{k=1}^N A_k (p_k)^n u[n]}_{\text{natural response}} + \underbrace{\sum_{k=1}^L Q_k (q_k)^n u[n]}_{\text{forced response}}$$

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- Influence of the input signal on the **natural response** is through the scale factor  $\{A_k\}$
- Influence of the system on the **forced response** is through the scale factor  $\{Q_k\}$

# Response with nonzero initial conditions

$$\sum_{k=0}^N a_k y[n-k] = \sum_{k=0}^M b_k x[n-k]$$

- $x[n]$  is assumed to be causal
- effects of all previous input signals are reflected in the initial conditions  $y[-1], y[-2], \dots, y[-N]$
- To determine  $y[n], \forall n \geq 0$ , the one-sided ZT can be used:

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$$Y^+(z) = \frac{\sum_{k=0}^M b_k z^{-k}}{1 + \sum_{k=1}^N a_k z^{-k}} X^+(z) - \frac{\sum_{k=1}^N a_k z^{-k} \sum_{n=1}^k y[-n] z^n}{1 + \sum_{k=1}^N a_k z^{-k}}$$

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- Since  $Y_{zi}^+(z) = \frac{N_0(z)}{A(z)}$  its poles are  $p_1, \dots, p_N$  and the zero-input response has the form:

$$y_{zi}[n] = \sum_{k=1}^N D_k(p_k)^n u[n]$$

# Response with nonzero initial conditions

- The terms involving the poles  $\{p_k\}$  can be combined:

$$y[n] = y_{zs}[n] + y_{zi}[n] = \sum_{k=1}^N A'_k (p_k)^n u[n] + \sum_{k=1}^L Q_k (q_k)^n u[n]$$

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# Schur-Cohn stability test

## Reminder

- LTI system  $\mathcal{H}$  is causal iff  $\mathcal{R}_H$  is the exterior of a circle (including  $\infty$ )
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Let be  $A(z) = 1 + a_1z^{-1} + a_2z^{-2} + \dots + a_Nz^{-N}$  the denominator polynomial of  $H(z)$ . A polynomial of degree  $m$  is denoted by:

$$A_m(z) = \sum_{k=0}^m a_m[k]z^{-k} \quad a_m(0) = 1$$

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The reverse polynomial  $B_m(z)$  of degree  $m$  is defined as:

$$B_m(z) = z^{-m} A_m(z^{-1}) = \sum_{k=0}^m a_m[m-k] z^{-k}$$

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3. Loop to step 2 until it fails or  $m = 1$

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- Define:

$$V_k(\omega) = |e^{j\omega} - z_k| \quad U_k(\omega) = |e^{j\omega} - p_k|$$
$$\theta_k(\omega) = \angle(e^{j\omega} - z_k) \quad \phi_k(\omega) = \angle(e^{j\omega} - p_k)$$

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$$V_k(\omega) = |e^{j\omega} - z_k| \quad U_k(\omega) = |e^{j\omega} - p_k|$$
$$\theta_k(\omega) = \angle(e^{j\omega} - z_k) \quad \phi_k(\omega) = \angle(e^{j\omega} - p_k)$$

- Magnitude response:

$$|H(\omega)| = |G| \frac{V_1(\omega) \cdots V_M(\omega)}{U_1(\omega) \cdots U_N(\omega)}$$

# Frequency response of rational systems

- Knowing poles and zeros of  $H(z)$  can be expressed as:

$$H(z) = Gz^{-K} \frac{\prod_{k=0}^M (z - z_k)}{\prod_{k=1}^N (z - p_k)}$$

- Frequency response:

$$H(\omega) = H(z)|_{z=e^{j\omega}} = Ge^{-j\omega K} \frac{\prod_{k=0}^M (j\omega - z_k)}{\prod_{k=1}^N (j\omega - p_k)}$$

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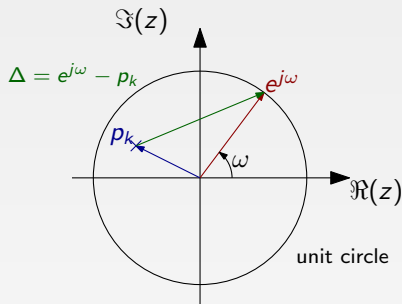
$$|H(\omega)| = |G| \frac{V_1(\omega) \cdots V_M(\omega)}{U_1(\omega) \cdots U_N(\omega)}$$

- Phase response:

$$\angle H(\omega) = \angle G - \omega K + \sum_{k=1}^{\infty} \theta_k(\omega) - \sum_{k=1}^N \phi_k(\omega)$$

# Geometrical interpretation

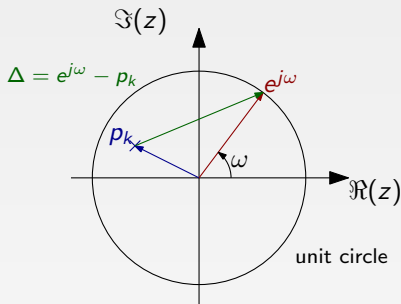
- Consider pole  $p_k$ :



- $\Delta = e^{j\omega} - p_k$ : vector joining  $p_k$  to point  $e^{j\omega}$  on unit circle
- $U_k(\omega) = |\Delta|$ : length of vector  $\Delta$
- $\phi_k(\omega) = \angle \Delta$ : angle between  $\Delta$  and real axis

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- $\phi_k(\omega) = \angle \Delta$ : angle between  $\Delta$  and real axis
- A similar interpretation holds for the terms  $V_k(\omega)$  and  $\theta_k(\omega)$  associated to the zeros  $z_k$

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- Zeros near the unit circle at  $z = re^{j\omega_0}$  give rise to:
  - deep notch in  $|H(\omega)|$  near  $\omega_0$
  - rapid phase variation near  $\omega_0$

## 6 Analysis of LTI systems in the z-Domain

- LTI systems described by LCCDE
- One-sided Z-Transform
- The system function
- Response of rational system Functions
- Schur-Cohn Stability test
- Frequency response of rational systems
- Analysis of certain basic systems

# First order LTI systems

The system function is given by:

$$H(z) = G \frac{1 - bz^{-1}}{1 - az^{-1}}$$

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  - causality: ROC:  $|z| > |a|$
  - stability:  $|a| < 1$
- Impulse response (ROC:  $|z| > |a|$ ):

$$h[n] = G \left( 1 - \frac{b}{a} \right) a^n u[n] + G \frac{b}{a} \delta[n]$$

# Low-pass case

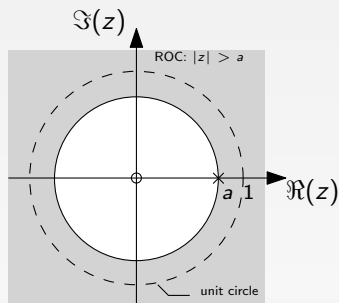
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- To get a low-pass behavior:  $a = 1 - \epsilon$ , where  $0 < \epsilon \ll 1$
- Additional attenuation of high-frequency is possible by proper placement of the zero  $z = b$ .

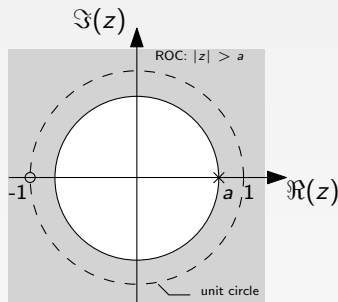
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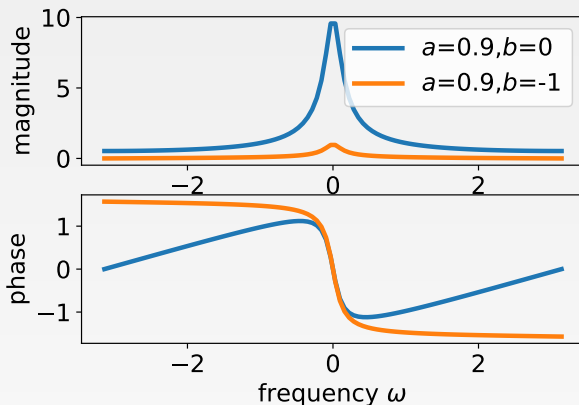


$$H_1(z) = G_1 \frac{1}{1 - az^{-1}} \quad (\text{zero: } b = 0)$$
$$G_1 = 1 - a \Rightarrow H_1(\omega = 0) = 1$$



$$H_2(z) = G_2 \frac{1 + z^{-1}}{1 - az^{-1}} \quad (\text{zero: } b = -1)$$
$$G_2 = \frac{1-a}{2} \Rightarrow H_2(\omega = 0) = 1$$

# Frequency responses of the corresponding low-pass systems





# High-pass case

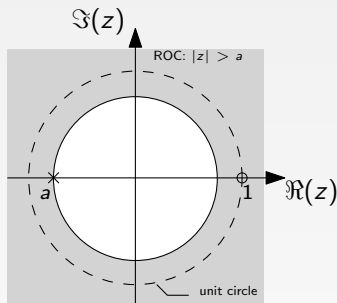
$$H(z) = G \frac{1 - bz^{-1}}{1 - az^{-1}}$$

- To get a high-pass behavior:  $a = -1 + \epsilon$ , where  $0 < \epsilon \ll 1$
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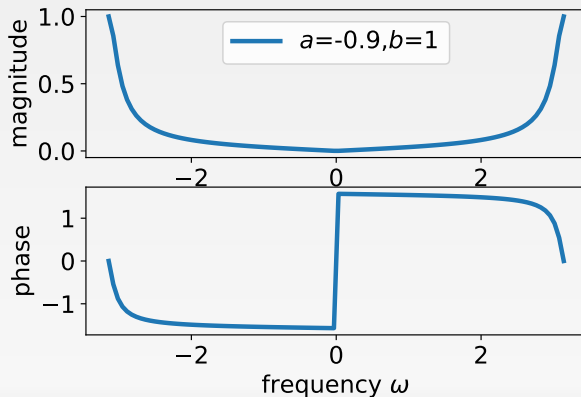


$$H_3(z) = G_3 \frac{1 - z^{-1}}{1 - az^{-1}} \quad (\text{zero: } b = 1)$$

$$G_3 = \frac{1+a}{2} \Rightarrow H_3(\omega = -\pi) = 1$$

# Frequency response of the corresponding high-pass system

- Frequency response:



# Second order systems

## System function

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### ■ Poles

- if  $a_1^2 > 4a_2$ : 2 distinct poles (real) at  $p_{1,2} = -\frac{a_1}{2} \pm \frac{1}{2}\sqrt{a_1^2 - 4a_2}$
- if  $a_1^2 = 4a_2$ : double pole (real) at  $p_1 = -\frac{a_1}{2}$
- if  $a_1^2 < 4a_2$ : 2 distinct poles (complex) at  $p_{1,2} = -\frac{a_1}{2} \pm j\frac{1}{2}\sqrt{4a_2 - a_1^2}$

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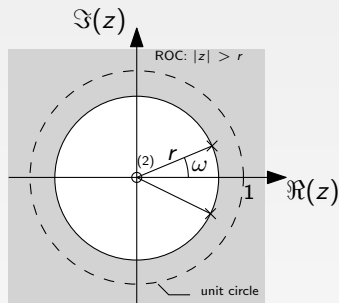
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### ■ Practical requirements:

- causality: ROC:  $|z| > \max\{|p_1|, |p_2|\}$
- stability:  $|p_1| < 1$  and  $|p_2| < 1 \Leftrightarrow |a_2| < 1$  and  $a_2 > |a_1| - 1$

# Second order systems: resonator

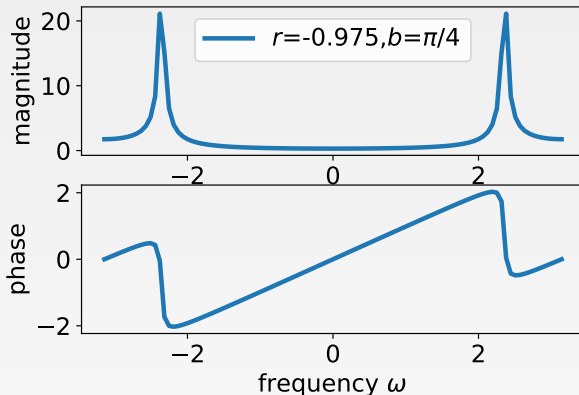


$$p_1 = re^{j\omega_0}$$
$$p_2 = re^{-j\omega_0} = p_1^*$$

$$H(z) = G \frac{1}{(1 - re^{j\omega_0}z^{-1})(1 - re^{-j\omega_0}z^{-1})}$$
$$= G \frac{1}{1 - 2r \cos(\omega_0) z^{-1} + r^2 z^{-2}}$$

# Second order systems: resonator

## ■ Frequency response:



- The frequency response clearly shows peaks around  $\pm\omega_0$ .
- For  $r$  close to 1 (but  $< 1$ ),  $|H(\omega)|$  reaches a maximum at  $\pm\omega_0$



## System function

$$\begin{aligned} H(z) &= B(z) = b_0 + b_1 z^{-1} + \dots + b_M z^{-M} \\ &= b_0 (1 - z_1 z^{-1}) \dots (1 - z_M z^{-1}) \end{aligned}$$

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- Impulse response:

$$h[n] = \begin{cases} b_n & 0 \leq n \leq M \\ 0 & \text{otherwise} \end{cases}$$

# FIR filters: moving average system

- Difference equation:

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



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- PZ analysis: roots of the numerator

$$z^M = 1 \Rightarrow z = e^{j2\pi k/M}, k = 0, 1, \dots, M-1$$

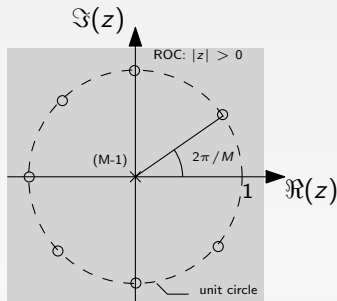
$\Rightarrow$  there is no pole at  $z = 1$  because of PZ cancellation:

$$H(z) = \frac{1}{M} \prod_{k=1}^{M-1} \left(1 - e^{j2\pi k/M} z^{-1}\right)$$

# FIR filters: moving average system

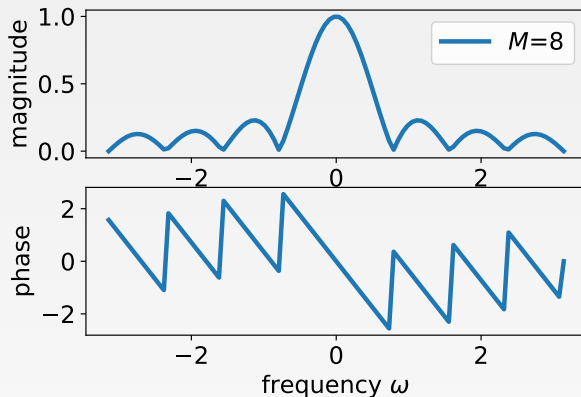
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- PZ diagram for  $M = 8$



# FIR filters: moving average system

- Frequency response:



- 1 Introduction
- 2 Discrete-Time signals
- 3 Discrete-Time systems
- 4 The Z-Transform (ZT)
- 5 Fourier Transform of DT signals
- 6 Analysis of LTI systems in the z-Domain
- 7 Discrete Fourier Transform (DFT)**

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Several drawbacks from a computational viewpoint:

- the summation over  $n$  is infinite
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Several drawbacks from a computational viewpoint:

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In many situation, it is either not possible, or not necessary to implement the infinite summation:

- only the signal samples  $x[n]$  from  $n$  to  $N - 1$  are available
- the signal is known to be zero outside this range; or
- the signal is periodic with period  $N$



## 7 Discrete Fourier Transform (DFT)

- The DFT and its inverse
- Relationship between the DFT and the DTFT
- Properties of the DFT
- Relation between linear and circular convolutions
- The FFT

# The DFT and its inverse

## Definition of the DFT

$$\begin{aligned} X[k] = \text{DFT}_N \{x[n]\} &\triangleq \sum_{n=0}^{N-1} x[n] e^{-j2\pi kn/N}, \quad k \in \mathbb{Z} \\ &\triangleq \sum_{n=0}^{N-1} x[n] e^{-j\omega_k n}, \quad \omega = 2\pi k/N \end{aligned}$$

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- The “D” in DFT stands for **discrete frequency** (i.e.  $\omega_k$ )

# Examples

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$$x[n] = \begin{cases} 1 & n = 0 \\ 0 & n = 1, \dots, N-1 \end{cases}$$

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case  $a = 1$

$$\Rightarrow X[k] = \begin{cases} N & \text{if } k = 0 \\ 0 & \text{if } k = 1, \dots, N-1 \end{cases}$$

case  $a = e^{j2\pi l/N}$

$$\Rightarrow X[k] = \begin{cases} N & \text{if } k = l \text{ modulo } N \\ 0 & \text{otherwise} \end{cases}$$

# Inverse DFT (IDFT)

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If  $X[k]$  is the  $N$ -point DFT of the samples  $\{x[0], \dots, x[N-1]\}$  then:

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## 7 Discrete Fourier Transform (DFT)

- The DFT and its inverse
- Relationship between the DFT and the DTFT
- Properties of the DFT
- Relation between linear and circular convolutions
- The FFT

# Introduction

The DFT may be viewed as a finite approximation to the DTFT:

$$X[k] = \sum_{n=0}^{N-1} x[n]e^{-j\omega_k n} \approx X(\omega_k = \frac{2\pi k}{N}) = \sum_{n=-\infty}^{\infty} x[n]e^{-j\omega_k n}$$

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- It should not be possible to recover the DTFT exactly from the DFT
  - an arbitrary signal  $x[n]$  cannot be recovered entirely from its  $N$ -point DFT
- However, in the following two special cases the DTFT can be evaluated exactly at any frequency  $\omega \in [-\pi, \pi]$  if the DFT is known:
  - **finite length signals**
  - **$N$ -periodic signals**

# Finite length signals

## Assumption

Suppose  $x[n] = 0$  for  $n < 0$  and for  $n \geq N$

## Inverse DFT

In this case  $x[n]$  can be recovered entirely from its  $N$ -point DFT:

$$\tilde{x}[n] = \text{IDFT} \{X[k]\} = \frac{1}{N} \sum_{k=0}^{N-1} X[k] e^{j2\pi kn/N}, \quad n \in \mathbb{Z}$$

- For  $n = 0, \dots, N-1$  the IDFT theorem yields:  $x[n] = \tilde{x}[n]$
- For  $n < 0$  and for  $n \geq N$ , by assumption:  $x[n] = 0$



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$$x[n] = \begin{cases} \tilde{x}[n] & \text{if } 0 \leq n \leq N \\ 0 & \text{otherwise} \end{cases}$$

# Finite length signals

## Relationship between DFT and DTFT

In this case the DTFT  $X(\omega = \omega_k = 2\pi k/N)$  can be completely reconstructed from the  $N$ -point DFT  $X[k]$ :

$$X(\omega_k) = \sum_{n=-\infty}^{\infty} x[n]e^{-j\omega_k n} = \sum_{n=0}^{N-1} x[n]e^{-j\omega_k n} = X[k]$$

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In the general case, i.e.  $\omega$  arbitrary, the following theorem can be applied.

### Theorem

$X(\omega)$  and  $X[k]$  respectively denote the DTFT and  $N$ -point DFT of signal  $x[n]$  ( $x[n] = 0$  for  $n < 0$  and for  $n \geq N$ ):

$$X(\omega) = \sum_{k=0}^{N-1} X[k]P(\omega - \omega_k)$$

where

$$P(\omega) \triangleq \frac{1}{N} \sum_{n=0}^{N-1} e^{-j\omega n}$$

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$$X(\omega) = \sum_{k=0}^{N-1} X[k]P(\omega - \omega_k) \text{ with } P(\omega) \triangleq \frac{1}{N} \sum_{n=0}^{N-1} e^{-j\omega n}$$

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$$P(\omega) = \frac{1}{N} \frac{1 - e^{-j\omega N}}{1 - e^{j\omega}} = \frac{1}{N} e^{-j\omega(N-1)/2} \frac{\sin(\omega N/2)}{\sin(\omega/2)}$$

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- Note that at frequency  $\omega_k = 2\pi/N$

$$P(\omega_k) = \begin{cases} 1 & k = 0 \\ 0 & k = 1, \dots, N-1 \end{cases}$$



# Periodic signals

## Assumption

Suppose  $x[n]$  is  $N$ -periodic, i.e.  $x[n + N] = x[n]$

## Inverse DFT

In this case  $x[n]$  can be recovered entirely from its  $N$ -point DFT:

$$\tilde{x}[n] = \text{IDFT} \{X[k]\} = \frac{1}{N} \sum_{k=0}^{N-1} X[k] e^{j2\pi kn/N}, n \in \mathbb{Z}$$

- For  $n = 0, \dots, N - 1$  the IDFT theorem yields:  $x[n] = \tilde{x}[n]$
- Since both  $x[n]$  and  $\tilde{x}[n]$  are known to be  $N$ -periodic, it follows that  $x[n] = \tilde{x}[n]$  must also be true for  $n < 0$  and for  $n \geq N$ :

$$x[n] = \tilde{x}[n], \forall n \in \mathbb{Z}$$

# Periodic signals

## Relationship between DFT and DTFT

Since the  $N$ -periodic signal  $x[n]$  can be recovered completely from its  $N$ -point DFT  $X[k]$ , it should be possible to reconstruct the DTFT  $X(\omega)$  from  $X[k]$ .

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$X(\omega)$  and  $X[k]$  respectively denote the DTFT and  $N$ -point DFT of signal  $x[n]$

$$X(\omega) = \frac{2\pi}{N} \sum_{-\infty}^{\infty} X[k] \delta_a(\omega - \omega_k)$$

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- $X(\omega) \Leftrightarrow$  periodic train of infinite impulses in the  $\omega$  domain
- When  $x[n]$  is  $N$ -periodic, the DFT admits a Fourier series interpretation since the IDFT provides an expansion of  $x[n]$  as a sum of harmonically related complex exponential signals  $e^{j\omega_k n}$ :

$$x[n] = \frac{1}{N} \sum_{k=0}^{N-1} X[k] e^{j\omega_k n}, \quad n \in \mathbb{Z}$$

# Signal reconstruction via DTFT sampling

1. Let  $X(\omega)$  be the DTFT of signal  $x[n]$ ,  $n \in \mathbb{Z}$ , that is:

$$X(\omega) = \sum_{n=-\infty}^{\infty} x[n]e^{-j\omega n}, \omega \in \mathbb{R}.$$

2. Consider the sampled values of  $X(\omega)$  at uniformly spaced frequencies  $\omega_k = 2\pi k/N$  for  $k = 0, \dots, N-1$ .
3. Suppose we compute the IDFT of the samples  $X(\omega_k)$ :

$$\hat{x}[n] = \text{IDFT} \{X(\omega_k)\} = \frac{1}{N} \sum_{k=0}^{N-1} X(\omega_k) e^{j\omega_k n}$$

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What is the relationship between the original signal  $x[n]$  and the reconstructed sequence  $\hat{x}[n]$  ?

- $\hat{x}[n]$  is  $N$ -periodic, while  $x[n]$  may not be
- Even for  $n = 0, \dots, N-1$  there is no reason for  $\hat{x}[n]$  to be equal to  $x[n]$

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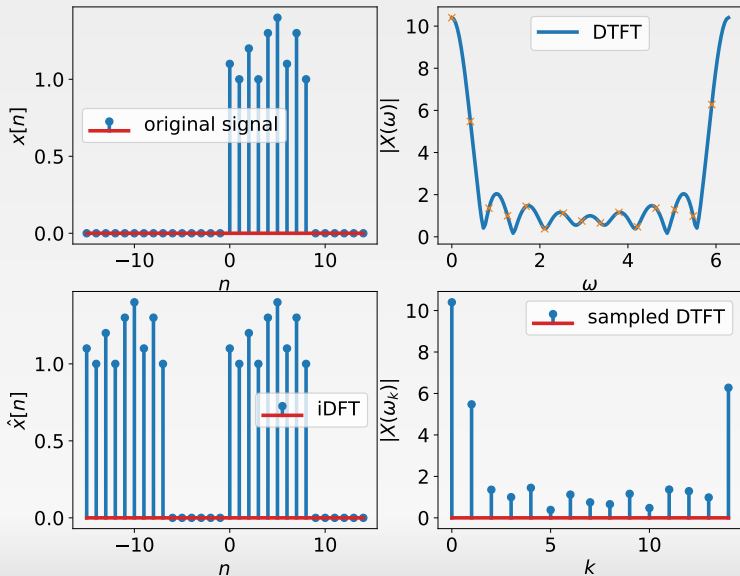
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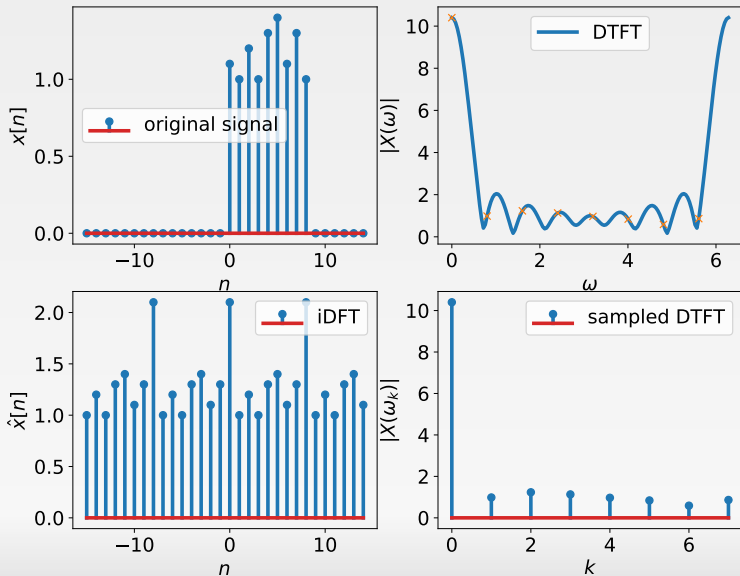
$$x[n] = \begin{cases} \hat{x}[n] & n = 0, \dots, N-1 \\ 0 & \text{otherwise} \end{cases}$$

2. **Non time-limited signal**: suppose that  $x[n] \neq 0$  for some  $n < 0$  or  $n \geq N$ . Then, the sequences  $x[n - rN]$  for different values of  $r$  will overlap in the time-domain. In this case, it is not true that  $\hat{x}[n] = x[n]$  for all  $0 \leq n \leq N-1 \Rightarrow$  **temporal aliasing**

# Time limited signal



# Non time-limited signal



## 7 Discrete Fourier Transform (DFT)

- The DFT and its inverse
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# Properties of the DFT

## Notations

$x[n]$  and  $y[n]$  are defined over  $0 \leq n \leq N - 1$ :

$$x[n] \stackrel{\text{DFT}_N}{\longleftrightarrow} X[k]$$

$$y[n] \stackrel{\text{DFT}_N}{\longleftrightarrow} Y[k]$$

$X[k]$  and  $Y[k]$  are viewed as  $N$ -periodic sequences, defined for all  $k \in \mathbb{Z}$ .



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## Modulo $N$ operation

any integer  $n \in \mathbb{Z}$  can be expressed uniquely as  $n = k + rN$  where  $k \in \{0, \dots, N - 1\}$  and  $r \in \mathbb{Z}$ :

$$(n)_N = n \text{ modulo } N \triangleq k$$

# Time reversal and complex conjugation

## Circular time reversal

Given a sequence  $x[n]$ ,  $0 \leq n \leq N - 1$ , its circular reversal (CR) is defined as:

$$\text{CR} \{x[n]\} = x[(-n)_N], 0 \leq n \leq N - 1$$

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Example: Let  $x[n] = 6 - n$  for  $n = 0, \dots, 5$ .

$n$	0	1	2	3	4	5
$x[n]$	6	5	4	3	2	1
$(-n)_6$	0	5	4	3	2	1
$x[(-n)_6]$	6	1	2	3	4	5

# Time reversal and complex conjugation

Interpretation:

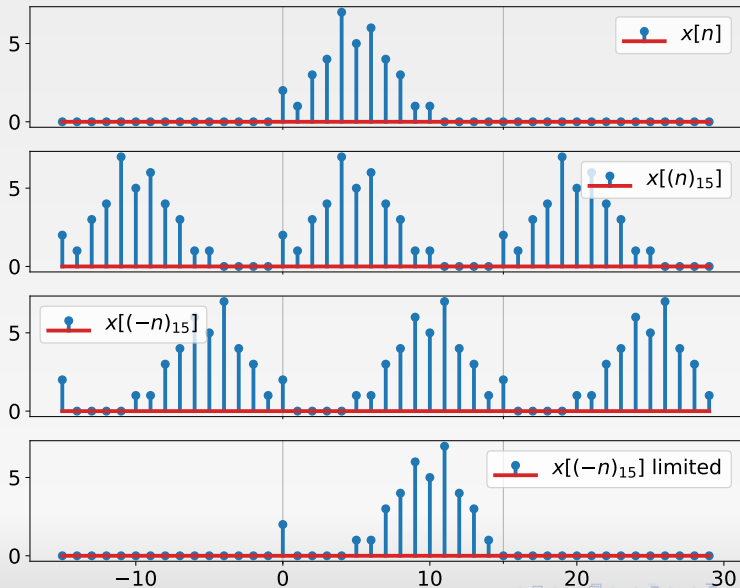
- Circular reversal can be seen as an operation on the set of samples  $x[0], \dots, x[N-1]$ :
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- One can also see this operation consisting in:
  1. periodizing the samples of  $x[n]$ ,  $0 \leq n \leq N - 1$  with period  $N$
  2. time-reversing the periodized sequence
  3. keeping only the samples between 0 and  $N - 1$

# Time reversal and complex conjugation



# Time reversal and complex conjugation

## Property

$$x[(-n)_N] \stackrel{\text{DFT}_N}{\longleftrightarrow} X[-k]$$

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# Time reversal and complex conjugation

## Property

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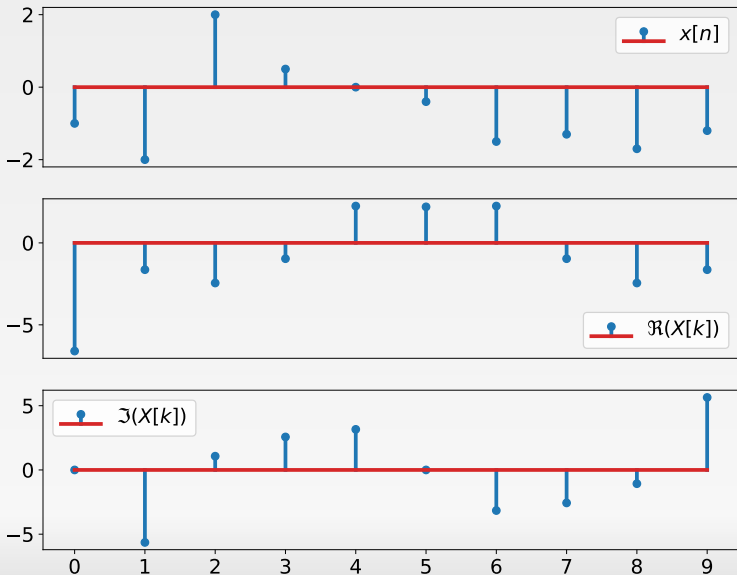
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  - $X[N - k] = X^*[k]$  for  $1 \leq k \leq N - 1$

# Time reversal and complex conjugation



## Linearity

$$ax[n] + by[n] \xrightarrow{\text{DFT}_N} aX[k] + bY[k]$$



# Even and odd decomposition

## Conjugate symmetric components of finite sequences

$$x_{e,N}[n] \triangleq \frac{1}{2} (x[n] + x^* [(-n)_N])$$

$$x_{o,N}[n] \triangleq \frac{1}{2} (x[n] - x^* [(-n)_N])$$

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$$j\Im \{x[n]\} \xleftrightarrow{\text{DFT}_N} X_o[k]$$

$$x_{e,N}[n] \xleftrightarrow{\text{DFT}_N} \Re \{X[k]\}$$

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Given a sequence  $x[n]$  defined over the interval  $0 \leq n \leq N - 1$ , we define its circular shift by  $k$  as follows:

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Example: Let  $x[n] = 6 - n$  for  $n = 0, \dots, 5$ .

$n$	0	1	2	3	4	5
$x[n]$	6	5	4	3	2	1
$(n - 2)_6$	4	5	0	1	2	3
$x[(n - 2)_6]$	2	1	6	5	4	3

Interpretation:

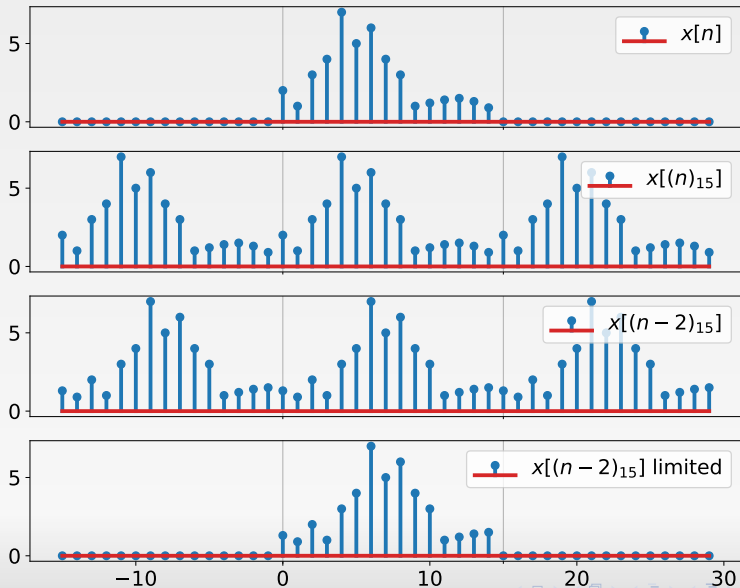
- Can be seen as an operation on the set of signal samples  $x[n]$  in which:
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  - any signal sample leaving the interval  $0 \leq n \leq N - 1$  from one end reenters by the other end

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  - any signal sample leaving the interval  $0 \leq n \leq N - 1$  from one end reenters by the other end
- Alternatively, it may be interpreted as follows:
  1. periodizing the samples of  $x[n]$ ,  $0 \leq n \leq N - 1$  with period  $N$
  2. delaying the periodized sequence by  $k$  samples
  3. keeping only the samples between 0 and  $N - 1$



# Circular shift



## Circular shift property

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- Since the DFT  $X[k]$  is already periodic, the modulo  $N$  operation is not needed here, that is:  $X[(k - m) - N] = X[k - m]$ .

# Circular convolution

## Definition

Let  $x[n]$  and  $y[n]$  be 2 sequences defined over  $0 \leq n \leq N - 1$ :

$$x[n] \circledast y[n] \triangleq \sum_{m=0}^{N-1} x[m]y[(n-m)_N], \quad 0 \leq n \leq N-1$$

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$$x[n] \circledast y[n] \xrightarrow{\text{DFT}_N} X[k] Y[k]$$

## Multiplication Property

$$x[n] y[n] \xrightarrow{\text{DFT}_N} \frac{1}{N} X[k] \circledast Y[k]$$

# Other properties

## Plancherel's relation

$$\sum_{n=0}^{N-1} x[n]y^*[n] = \frac{1}{N} \sum_{k=0}^{N-1} X[k]Y^*[k]$$

## Parseval's relation

$$\sum_{n=0}^{N-1} |x[n]|^2 = \frac{1}{N} \sum_{k=0}^{N-1} |X[k]|^2$$



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- Parseval's relation is a special case of Plancherel's relation: with  $y[n] = x[n]$
- It allows the computation of the energy of the signal samples  $x[n]$  ( $n = 0, \dots, N - 1$ ) directly from the DFT samples  $X[k]$

## 7 Discrete Fourier Transform (DFT)

- The DFT and its inverse
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# Introduction

## Linear convolution

Time domain expression:

$$y_I[n] = x_1[n] * x_2[n] = \sum_{k=-\infty}^{\infty} x_1[k]x_2[n-k], n \in \mathbb{Z}$$

Frequency domain representation via DTFT:

$$Y_I(\omega) = X_1(\omega)X_2(\omega), \omega \in [0, 2\pi]$$

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## Circular convolution

Time domain expression:

$$y_c[n] = x_1[n] \circledast x_2[n] = \sum_{k=0}^{N-1} x_1[k]x_2[(n-k)_N], 0 \leq n \leq N-1$$

Frequency domain representation via  $N$ -point DFT

$$Y_c[k] = X_1[k]X_2[k], k \in \{0, \dots, N-1\}$$

# A necessary condition...

Circular convolution and linear convolution are equivalent if:

$$y_l[n] = \begin{cases} y_c[n] & \text{if } 0 \leq n \leq N \\ 0 & \text{otherwise} \end{cases}$$

$\Rightarrow$  true if signals  $x_1[n]$  and  $x_2[n]$  have both finite length.

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## Finite length assumption

Suppose that  $x_1[n]$  and  $x_2[n]$  are time limited to  $0 \leq n < N_1$  and  $0 \leq n < N_2$  respectively then the linear convolution is **time limited to**  
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Example: Consider  $x_1[n] = \{1, 1, 1, 1\}$  and  $x_2[n] = \{1, 1/2, 1/2\}$

- $N_1 = 4$  and  $N_2 = 3$
- $y_l[n] = \{1, 1.5, 2, 2, 1, .5\}$
- $\Rightarrow N_3 = 6 = N_1 + N_2 - 1$

# ...proved to be a sufficient condition

Assuming  $N \geq N_1 + N_2 - 1$ :

1. Linear convolution gives:

$$y_l[n] = \sum_{k=-\infty}^{\infty} x_1[k]x_2[n-k] = \sum_{k=0}^n x_1[k]x_2[n-k], \quad 0 \leq n < N$$



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## Conclusion

The linear and circular convolution are equivalent if and only if:

$$N \geq N_1 + N_2 - 1$$

# Relationship between $y_c[n]$ and $y_l[n]$

Assuming that  $N \geq \max\{N_1, N_2\}$  the DFT of the 2 sequences  $x_1[n]$  and  $x_2[n]$  are samples of the corresponding DTFT:

$$N \geq N_1 \Rightarrow X_1[k] = X_1(\omega_k), \omega_k = 2\pi k/N$$

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$\Rightarrow$  The circular convolution  $y_c[n]$  can be computed as the  $N$ -point IDFT of these frequency samples:

$$y_c[n] = \text{IDFT}_N \{Y_c[k]\} = \text{IDFT}_N \{Y_l(\omega_k)\}$$

# Relationship between $y_c[n]$ and $y_l[n]$

Applying the “signal reconstruction via DTFT sampling” theorem we obtain:

$$y_c[n] = \sum_{r=-\infty}^{\infty} y_l[n - rN], \quad 0 \leq n < N$$

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To get  $y_c[n] = y_l[n]$  for  $0 \leq n < N$ , temporal aliasing must be avoided: the length of DFT  $\geq$  length of  $y_l[n]$ , i.e. :

$$N \geq N_1 + N_2 - 1$$



# Linear convolution via DFT

Linear convolution via DFT can be summarized according to the following steps:

- Suppose that  $x_1[n]$  and  $x_2[n]$  are time limited to  $0 \leq n < N_1$  and  $0 \leq n < N_2$  respectively

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- Compute the DFTs:

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$\Rightarrow$  algorithm complexity =  $N^2$

# The FFT

## Fast Fourier Transform

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- Assuming  $N = 2^k$
- Considering even and odd part of the signals the  $\text{DFT}_N$  is split into 2  $\text{DFT}_{N/2}$
- The FFT leads to:
  - $\frac{N}{2} \log_2 N$  complex multiplications
  - $N \log_2 N$  complex additions
- The algorithm complexity becomes  $\frac{N}{2} \log_2 N$