

# Basics of Signals and Systems

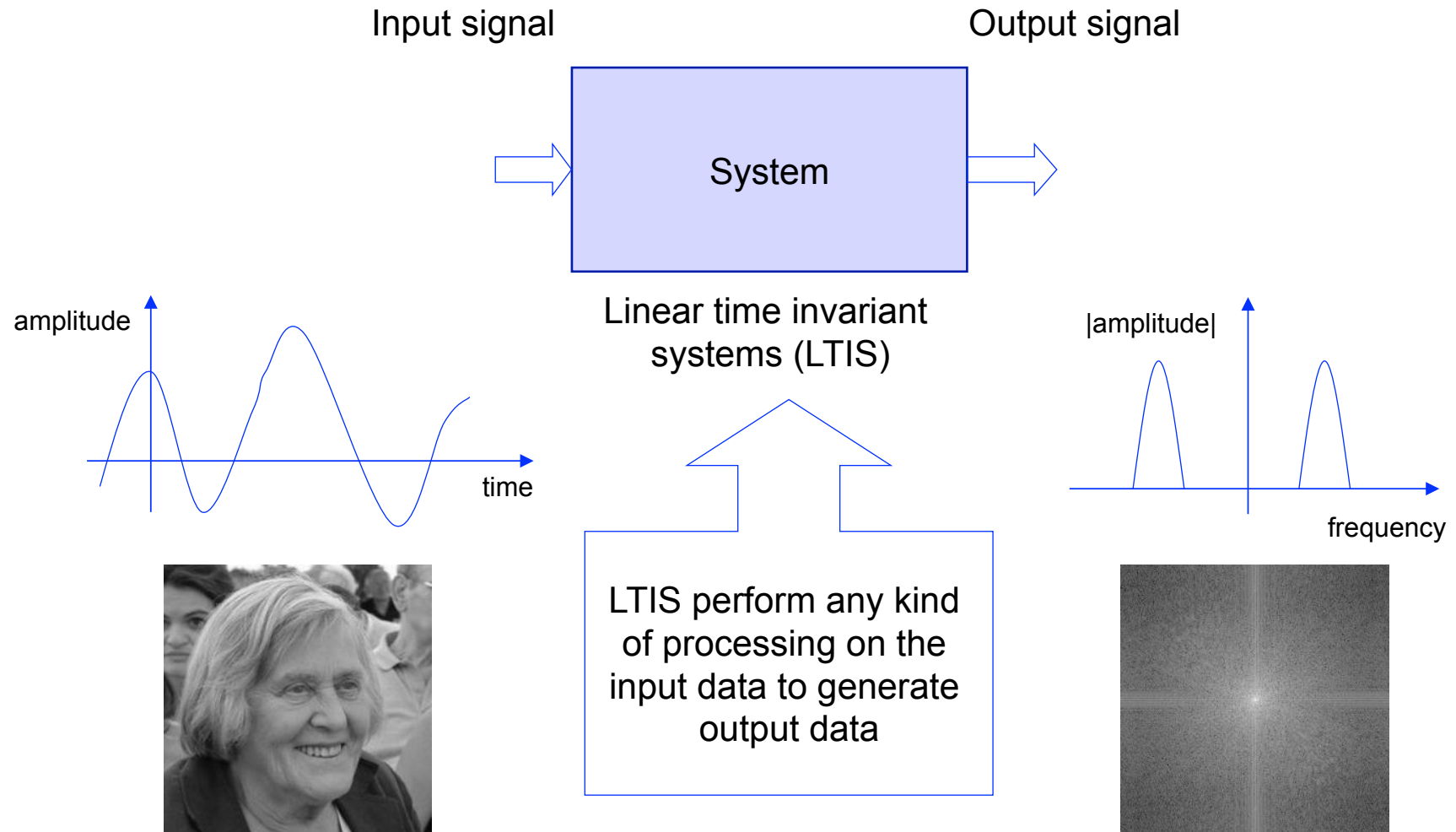
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AA 2015-2016

# Didactic material

- Textbook
  - Signal Processing and Linear Systems, B.P. Lathi, CRC Press
- Other books
  - Signals and Systems, Richard Baraniuk's lecture notes, available on line
  - Digital Signal Processing (4th Edition) (Hardcover), John G. Proakis, Dimitris K Manolakis
  - Teoria dei segnali analogici, M. Luise, G.M. Vitetta, A.A. D' Amico, McGraw-Hill
  - Signal processing and linear systems, Schaun's outline of digital signal processing
- All textbooks are available at the library
- Handwritten notes will be available on demand

# Signals&Systems



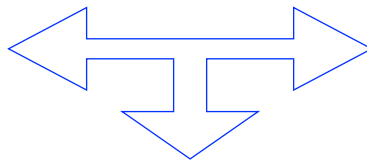
# Contents

## Signals

- Signal classification and representation
  - Types of signals
  - Sampling theory
  - Quantization
- Signal analysis
  - Fourier Transform
    - Continuous time, Fourier series, Discrete Time Fourier Transforms, Windowed FT
  - Spectral Analysis

## Systems

- Linear Time-Invariant Systems
  - Time and frequency domain analysis
  - Impulse response
  - Stability criteria
- Digital filters
  - Finite Impulse Response (FIR)
- Mathematical tools
  - Laplace Transform
    - Basics
  - Z-Transform
    - Basics

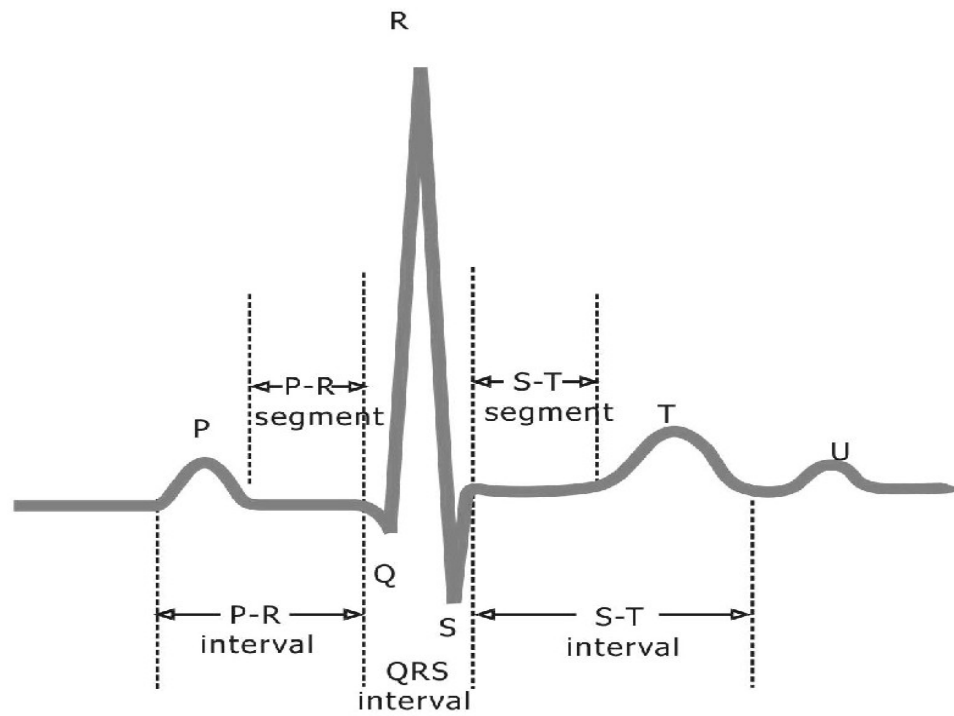


Applications in the domain of Bioinformatics

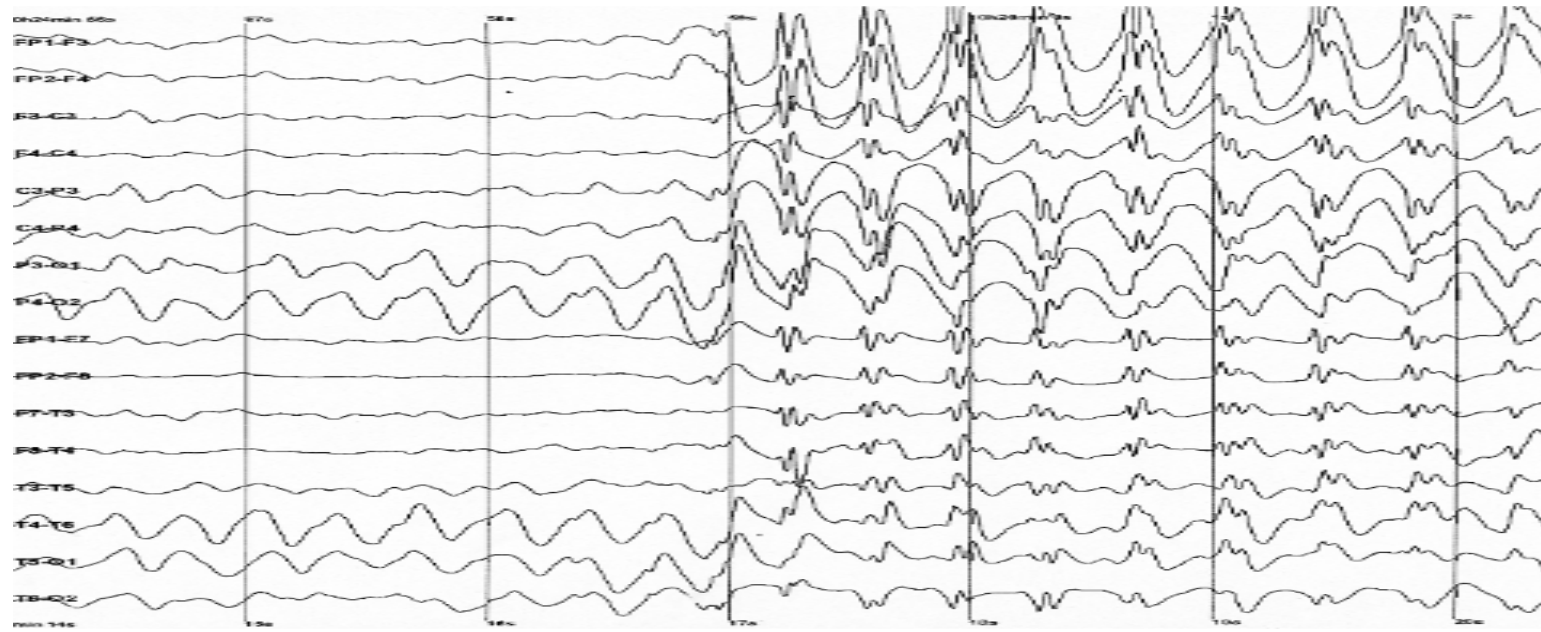
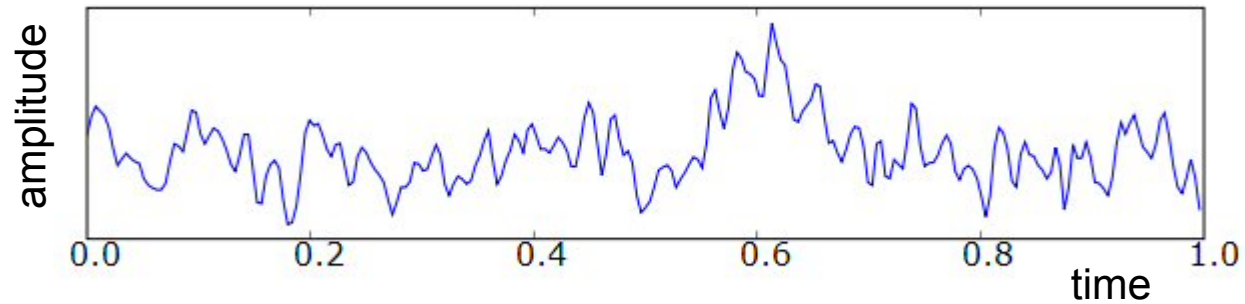
# What is a signal?

- A signal is a set of information of data
  - Any kind of physical variable subject to variations represents a signal
  - Both the independent variable and the physical variable can be either scalars or vectors
    - Independent variable: time ( $t$ ), space ( $x$ ,  $\mathbf{x}=[x_1, x_2]$ ,  $\mathbf{x}=[x_1, x_2, x_3]$ )
    - Signal:
      - Electrocardiography signal (EEG) 1D, voice 1D, music 1D
      - Images (2D), video sequences (2D+time), volumetric data (3D)

# Example: 1D biological signals: ECG

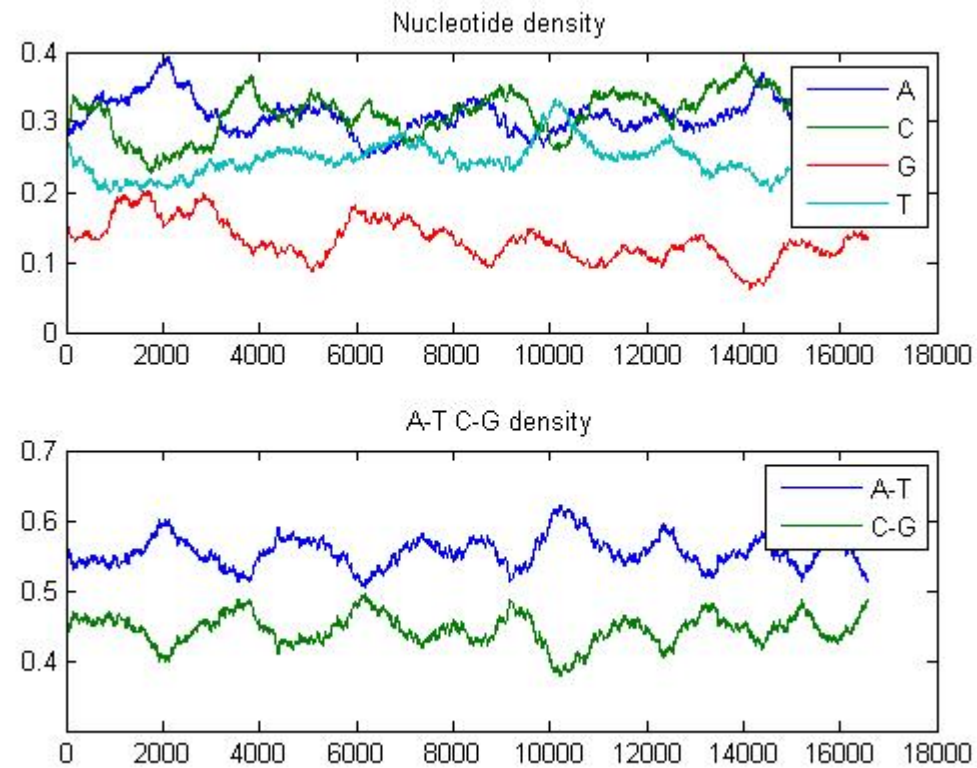


## Example: 1D biological signals: EEG



# 1D biological signals: DNA sequencing

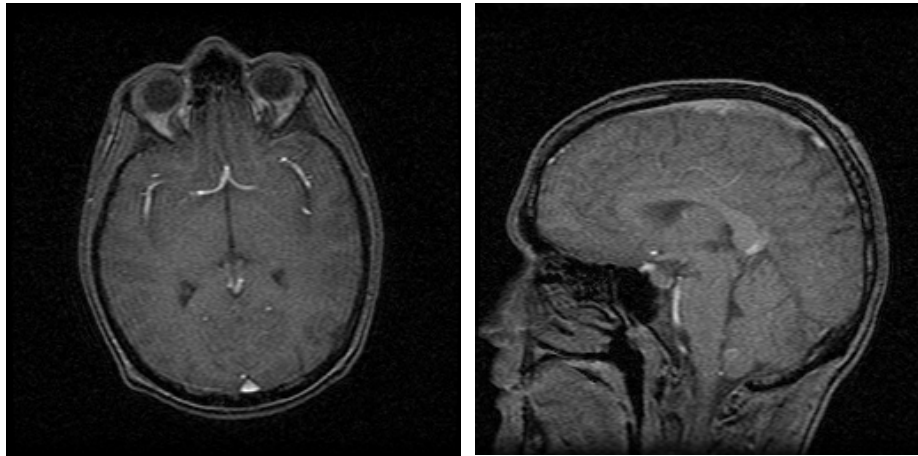
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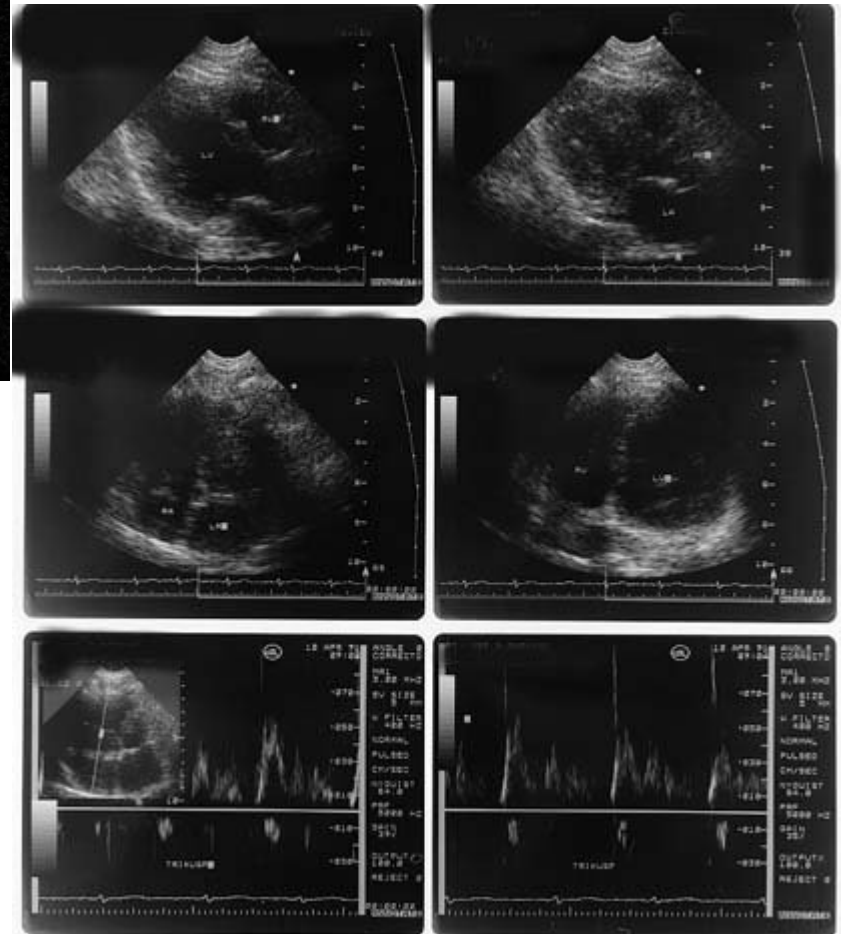


# Example: 2D biological signals: MI

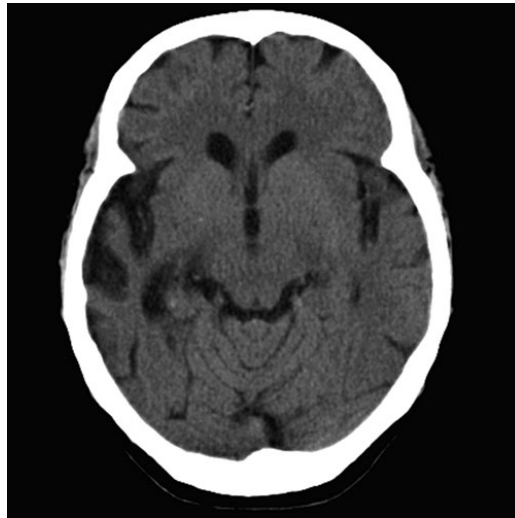
MRI



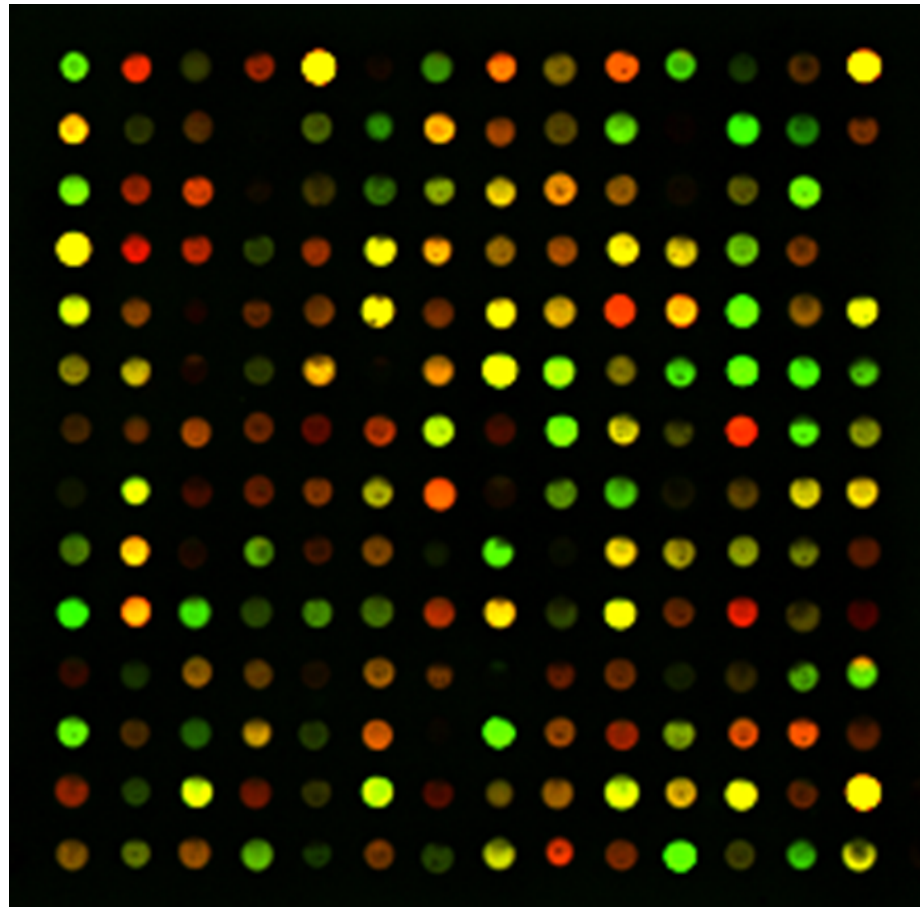
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## Example: 2D biological signals: microarrays



# Signals as functions

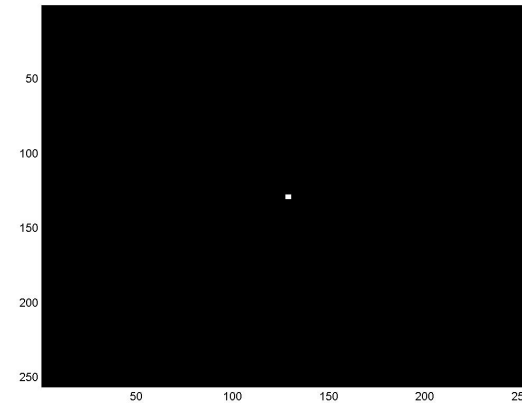
- Continuous functions of real independent variables
  - 1D:  $f=f(x)$
  - 2D:  $f=f(x,y)$   $x,y$
  - Real world signals (audio, ECG, images)
- Real valued functions of discrete variables
  - 1D:  $f=f[k]$
  - 2D:  $f=f[i,j]$
  - *Sampled* signals
- Discrete functions of discrete variables
  - 1D:  $f^d=f^d[k]$
  - 2D:  $f^d=f^d[i,j]$
  - *Sampled and quantized* signals

# Images as functions

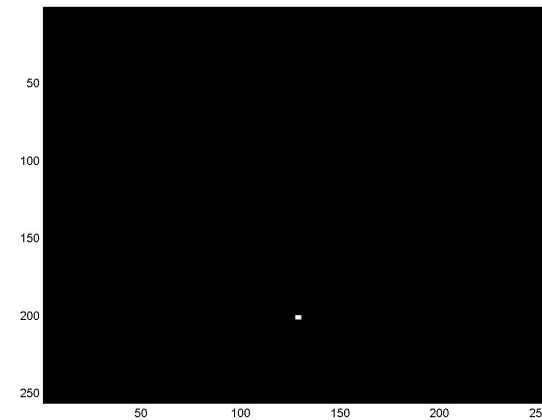
- Gray scale images: 2D functions
  - Domain of the functions: set of  $(x,y)$  values for which  $f(x,y)$  is defined : 2D lattice  $[i,j]$  defining the pixel locations
  - Set of values taken by the function : gray levels
- Digital images can be seen as functions defined over a discrete domain  $\{i,j: 0 < i < I, 0 < j < J\}$ 
  - $I,J$ : number of rows (columns) of the matrix corresponding to the image
  - $f=f[i,j]$ : gray level in position  $[i,j]$

## Example 1: $\delta$ function

$$\delta[i, j] = \begin{cases} 1 & i = j = 0 \\ 0 & i, j \neq 0; i \neq j \end{cases}$$



$$\delta[i, j - J] = \begin{cases} 1 & i = 0; j = J \\ 0 & \textit{otherwise} \end{cases}$$



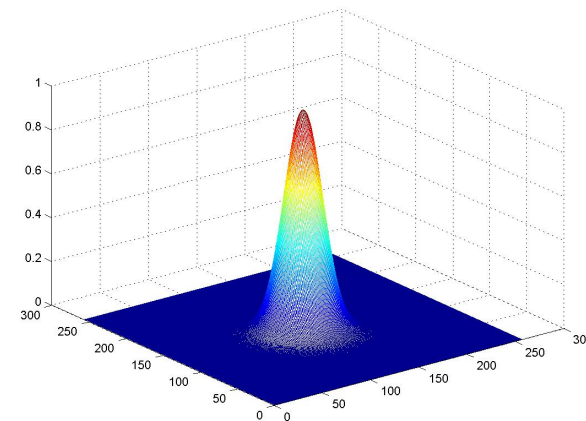
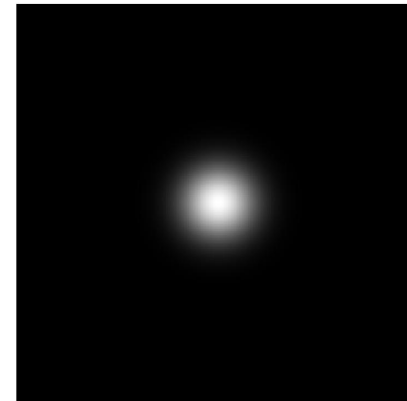
## Example 2: Gaussian

Continuous function

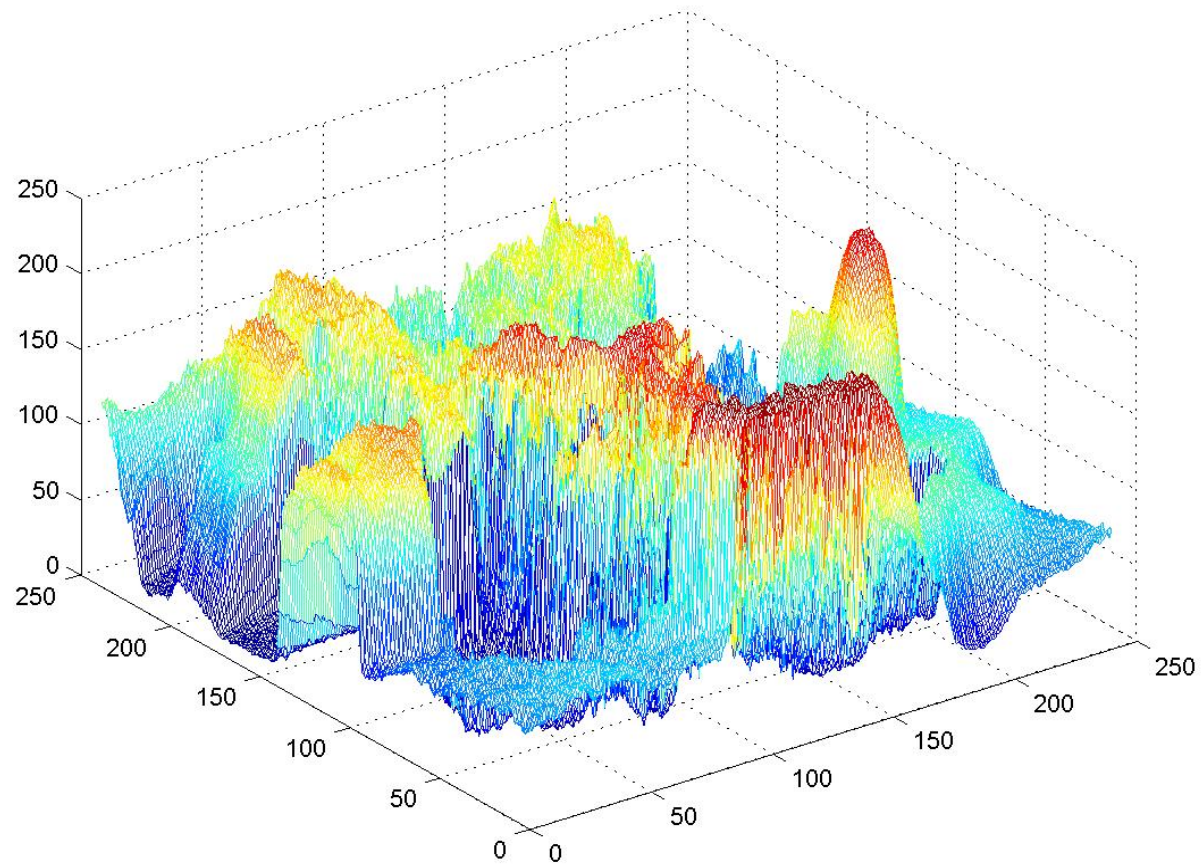
$$f(x, y) = \frac{1}{\sigma\sqrt{2\pi}} e^{-\frac{x^2+y^2}{2\sigma^2}}$$

Discrete version

$$f[i, j] = \frac{1}{\sigma\sqrt{2\pi}} e^{-\frac{i^2+j^2}{2\sigma^2}}$$



## Example 3: Natural image



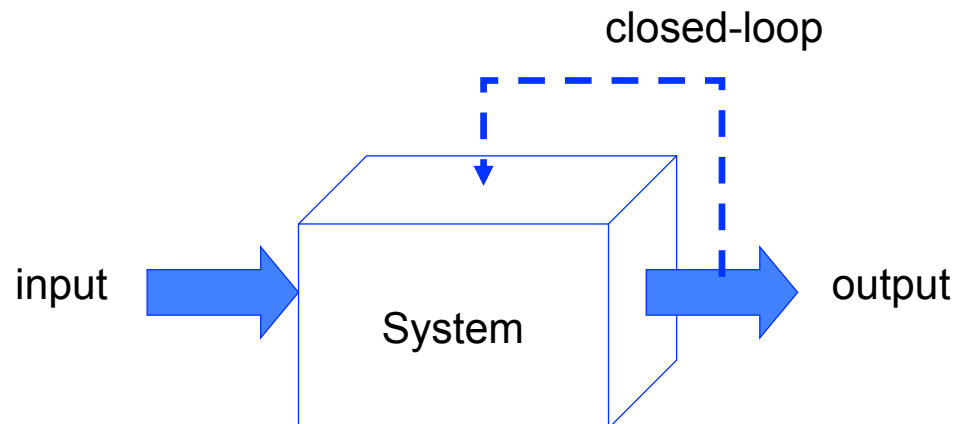
## Example 3: Natural image





# What is a system?

- Systems process signals to
  - Extract information (DNA sequence analysis)
  - Enable transmission over channels with limited capacity (JPEG, JPEG2000, MPEG coding)
  - Improve security over networks (encryption, watermarking)
  - Support the formulation of diagnosis and treatment planning (medical imaging)
  - .....



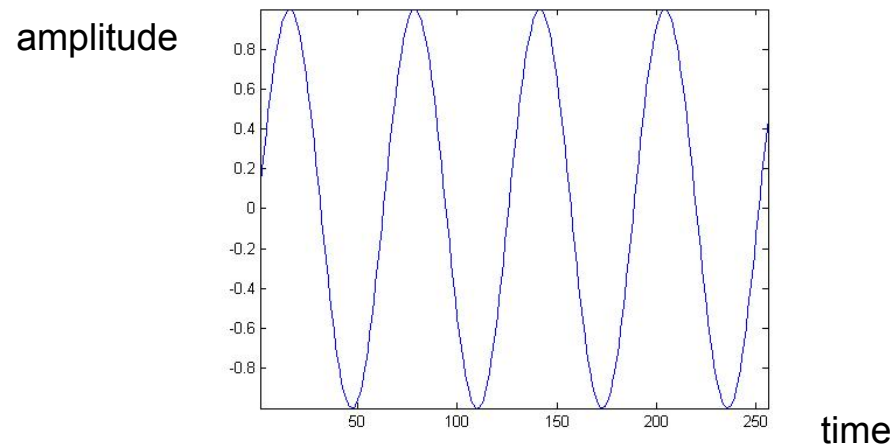
The function linking the output of the system with the input signal is called **transfer function** and it is typically indicated with the symbol  $h(\bullet)$

# Classification of signals

- Continuous time – Discrete time
- Analog – Digital (numerical)
- Periodic – Aperiodic
- Energy – Power
- Deterministic – Random (probabilistic)
- Note
  - Such classes are not disjoint, so there are digital signals that are periodic of power type and others that are aperiodic of power type etc.
  - Any combination of single features from the different classes is possible

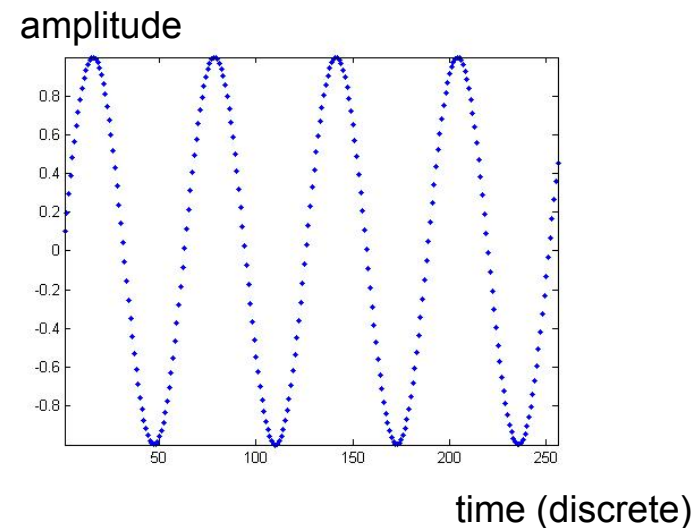
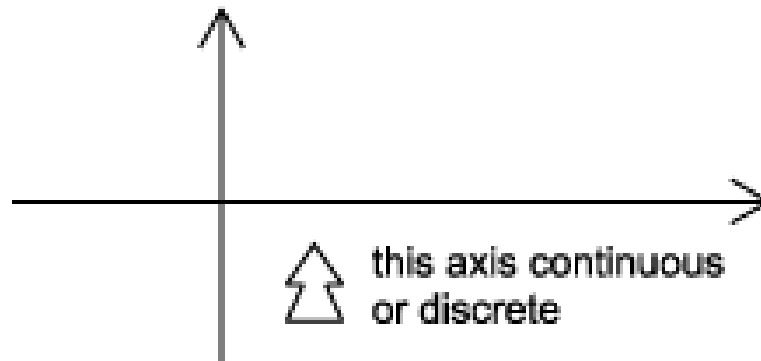
# Continuous time – discrete time

- Continuous time signal: a signal that is specified for every real value of the independent variable
  - The independent variable is continuous, that is it takes any value on the real axis
  - The domain of the function representing the signal has the cardinality of real numbers
    - Signal  $\leftrightarrow f=f(t)$
    - Independent variable  $\leftrightarrow$  time (t), position (x)
    - For continuous-time signals:  $t \in \mathbb{R}$



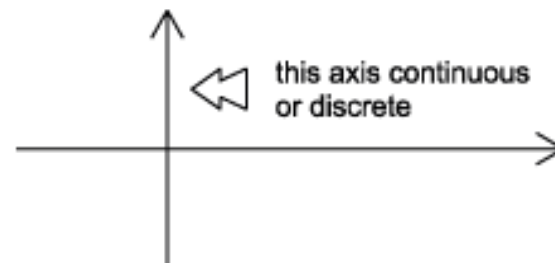
# Continuous time – discrete time

- Discrete time signal: a signal that is specified only for *discrete values* of the independent variable
  - It is usually generated by *sampling* so it will only have values at *equally spaced* intervals along the time axis
  - The domain of the function representing the signal has the cardinality of integer numbers
    - Signal  $\leftrightarrow f=f[n]$ , also called “sequence”
    - Independent variable  $\leftrightarrow n$
    - For discrete-time functions:  $t \in \mathbf{Z}$



# Analog - Digital

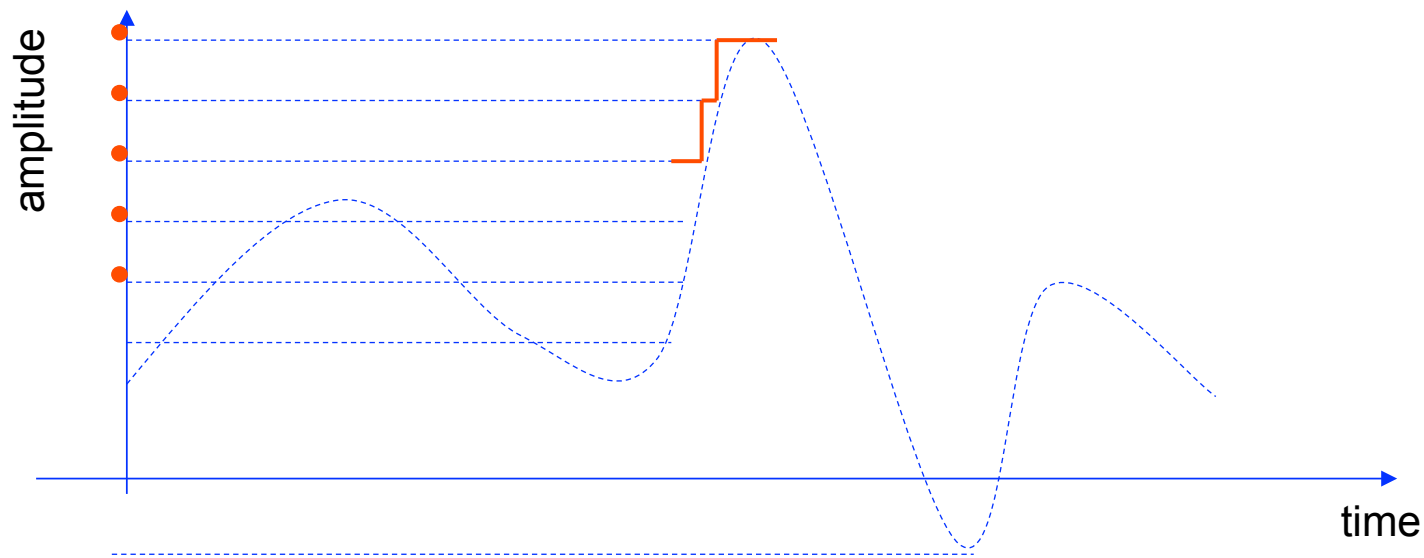
- **Analog signal:** signal whose amplitude can take on any value in a continuous range
  - The amplitude of the function  $f(t)$  (or  $f(x)$ ) has the cardinality of real numbers
    - The difference between analog and digital is similar to the difference between continuous-time and discrete-time. In this case, however, the difference is with respect to the value of the function (y-axis)
  - Analog corresponds to a continuous y-axis, while digital corresponds to a discrete y-axis



- *Here we call digital what we have called quantized in the EI class*
- *An analog signal can be both continuous time and discrete time*

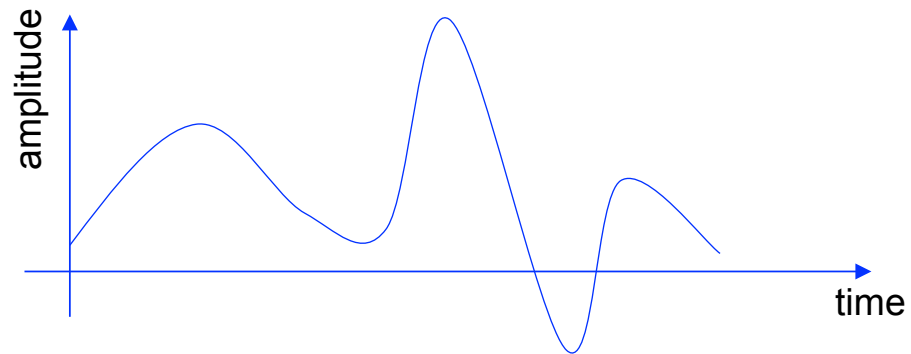
# Analog - Digital

- **Digital signal:** a signal is one whose amplitude can take on only a finite number of values (thus it is quantized)
  - The amplitude of the function  $f()$  can take only a finite number of values
  - A digital signal whose amplitude can take only  $M$  different values is said to be  $M$ -ary
    - Binary signals are a special case for  $M=2$



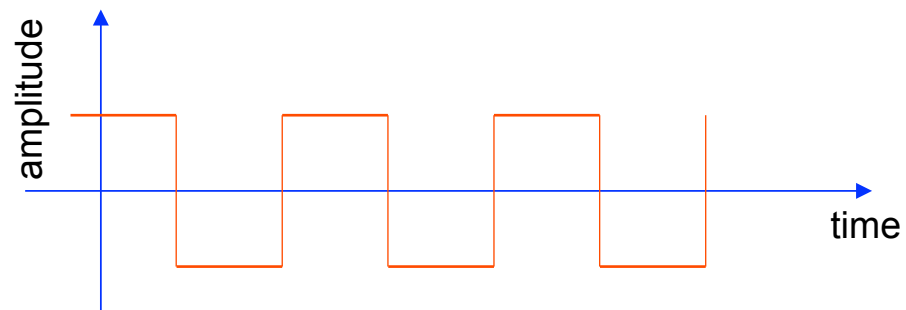
# Example

- Continuous time analog



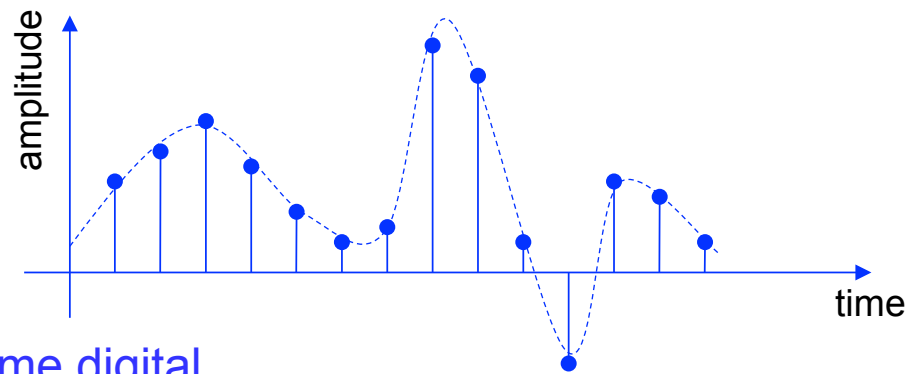
- Continuous time digital (or quantized)

- binary sequence, where the values of the function can only be one or zero.



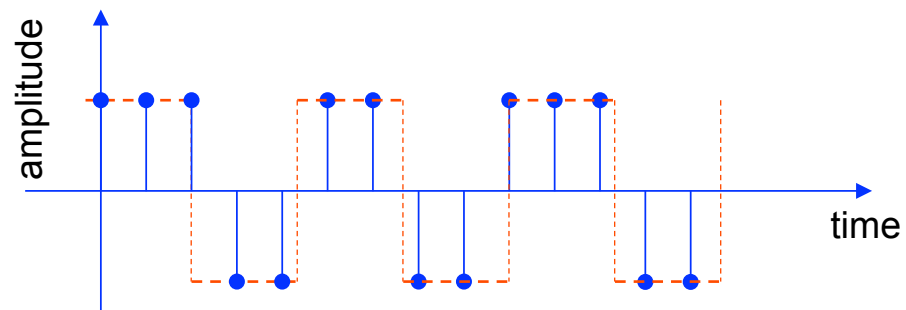
# Example

- Discrete time analog



- Discrete time digital

- binary sequence, where the values of the function can only be one or zero.





# Summary

Signal amplitude/ Time or space	Real	Integer
Real	Analog Continuous-time	Digital Continuous-time
Integer	Analog Discrete-time	Digital Discrete time

## Note

- In the image processing class we have defined as digital those signals that are both quantized and discrete time. It is a more restricted definition.
- The definition used here is as in the Lathi book.

# Periodic - Aperiodic

- A signal  $f(t)$  is *periodic* if there exists a positive constant  $T_0$  such that

$$f(t + T_0) = f(t) \quad \forall t$$

- The *smallest* value of  $T_0$  which satisfies such relation is said the *period* of the function  $f(t)$
- A periodic signal remains unchanged when *time-shifted* of integer multiples of the period
- Therefore, by definition, it starts at minus infinity and lasts forever

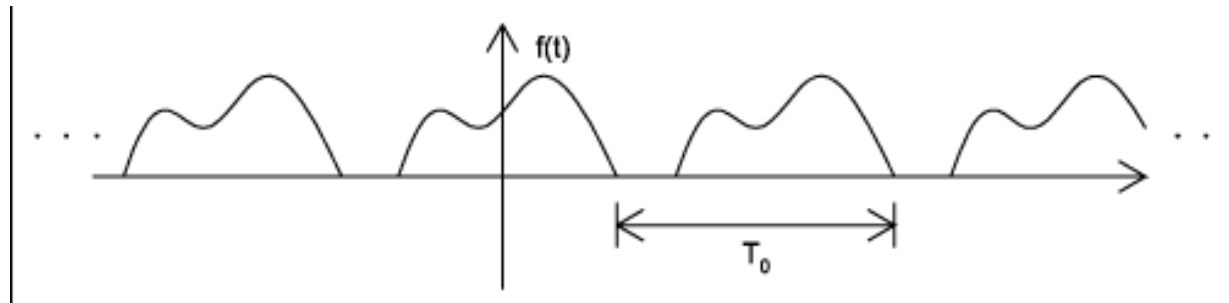
$$-\infty \leq t \leq +\infty \quad t \in \mathbb{R}$$

$$-\infty \leq n \leq +\infty \quad n \in \mathbb{Z}$$

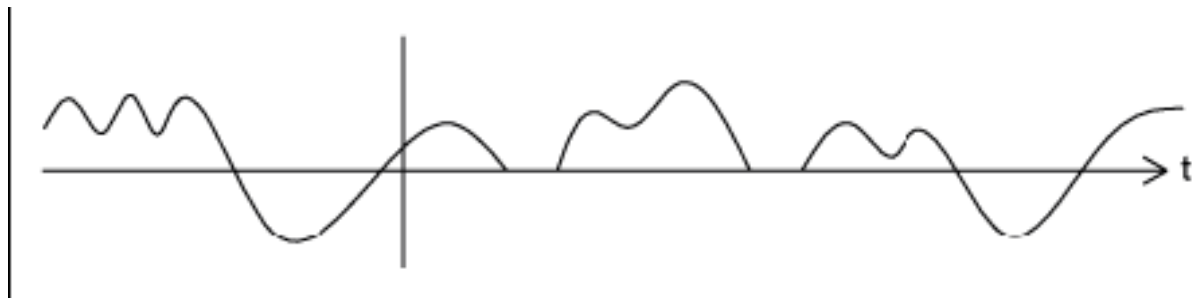
- Periodic signals can be generated by *periodical extension*

# Examples

- Periodic signal with period  $T_0$

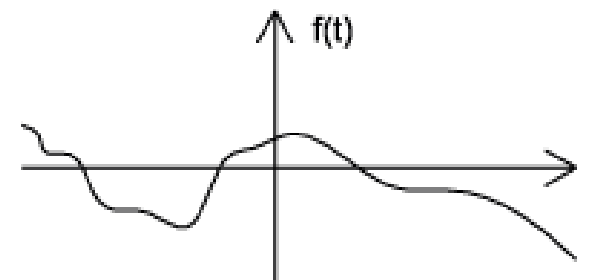
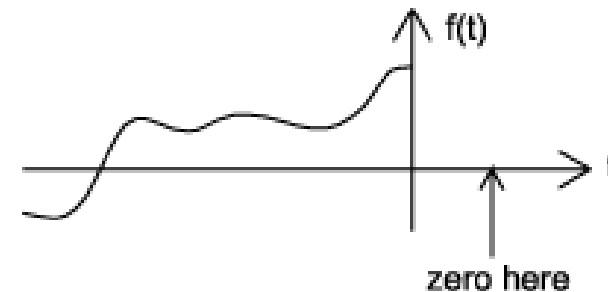
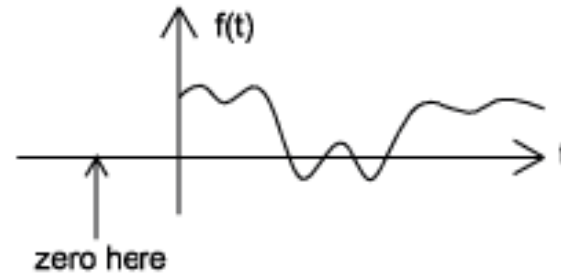


- Aperiodic signal



# Causal and non-Causal signals

- *Causal* signals are signals that are zero for all negative time (or spatial positions), while
- *Anticausal* are signals that are zero for all positive time (or spatial positions).
- *Noncausal* signals are signals that have nonzero values in both positive and negative time



# Causal and non-causal signals

- Causal signals

$$f(t) = 0 \quad t < 0$$

- Anticausals signals

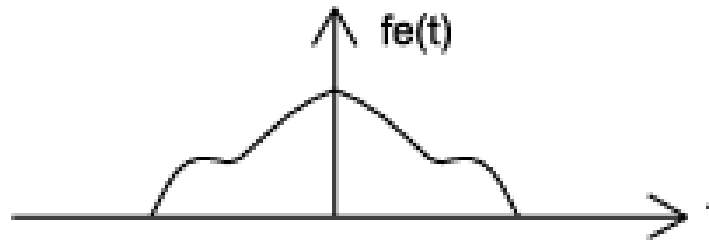
$$f(t) = 0 \quad t \geq 0$$

- Non-causal signals

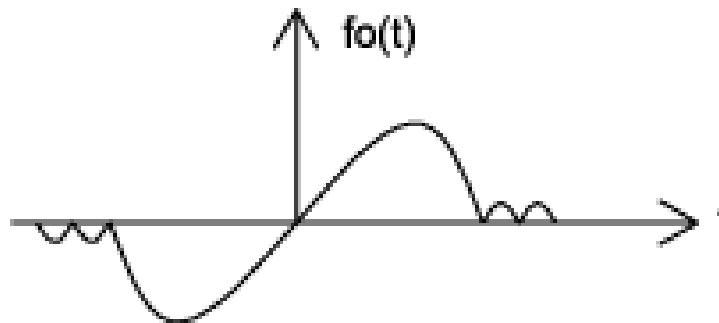
$$\exists t_1 < 0: \quad f(t_1) \neq 0$$

## Even and Odd signals

- An even signal is any signal  $f$  such that  $f(t) = f(-t)$ . Even signals can be easily spotted as they are symmetric around the vertical axis.



- An odd signal, on the other hand, is a signal  $f$  such that  $f(t) = -f(-t)$



## Decomposition in even and odd components

- Any signal can be written as a combination of an even and an odd signals
  - Even and odd components

$$f(t) = \frac{1}{2}(f(t) + f(-t)) + \frac{1}{2}(f(t) - f(-t))$$

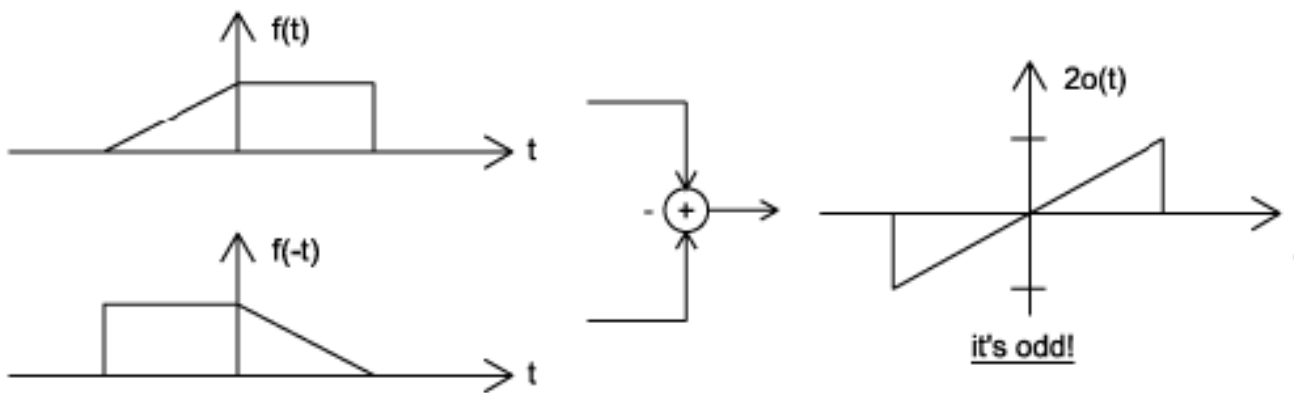
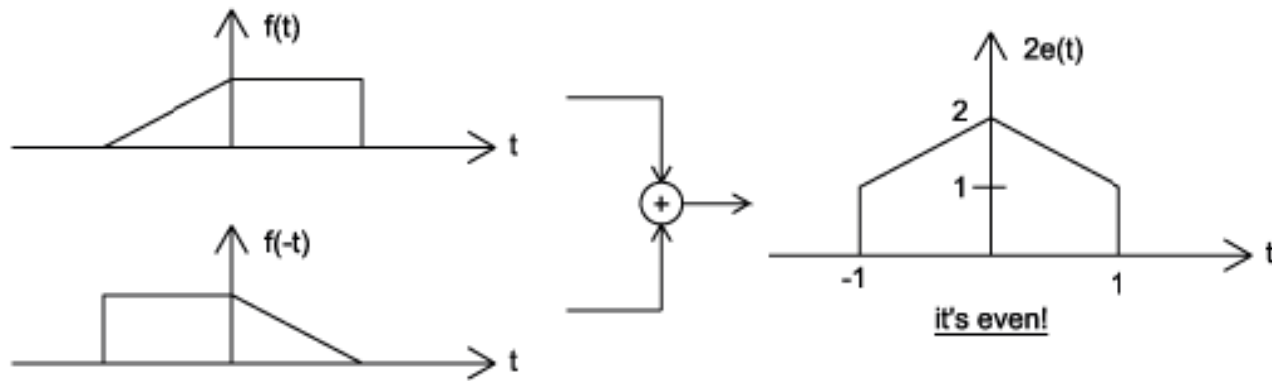
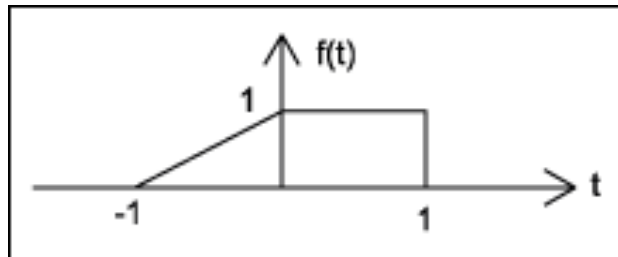
$$f_e(t) = \frac{1}{2}(f(t) + f(-t)) \quad \text{even component}$$

$$f_o(t) = \frac{1}{2}(f(t) - f(-t)) \quad \text{odd component}$$

$$f(t) = f_e(t) + f_o(t)$$

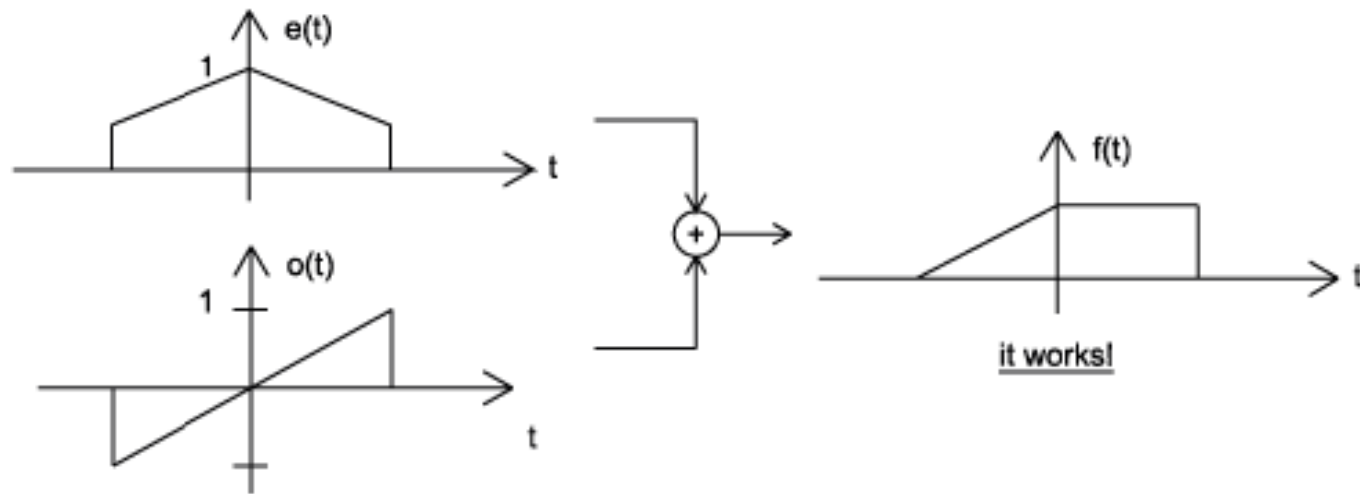


# Example



# Example

- Proof



## Some properties of even and odd functions

- even function x odd function = odd function
- odd function x odd function = even function
- even function x even function = even function
- Area

$$\int_{-a}^a f_e(t) dt = 2 \int_0^a f_e(t) dt$$

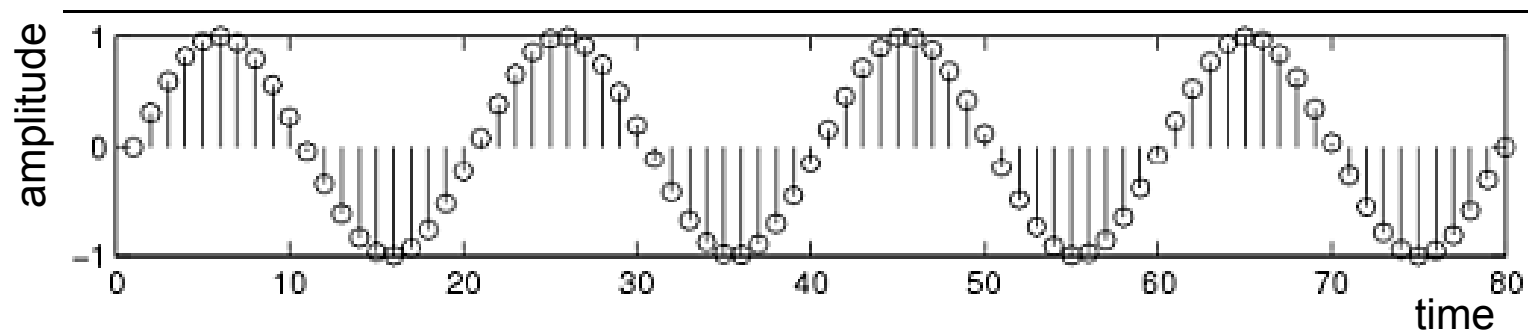
$$\int_{-a}^a f_o(t) dt = 0$$

# Deterministic - Probabilistic

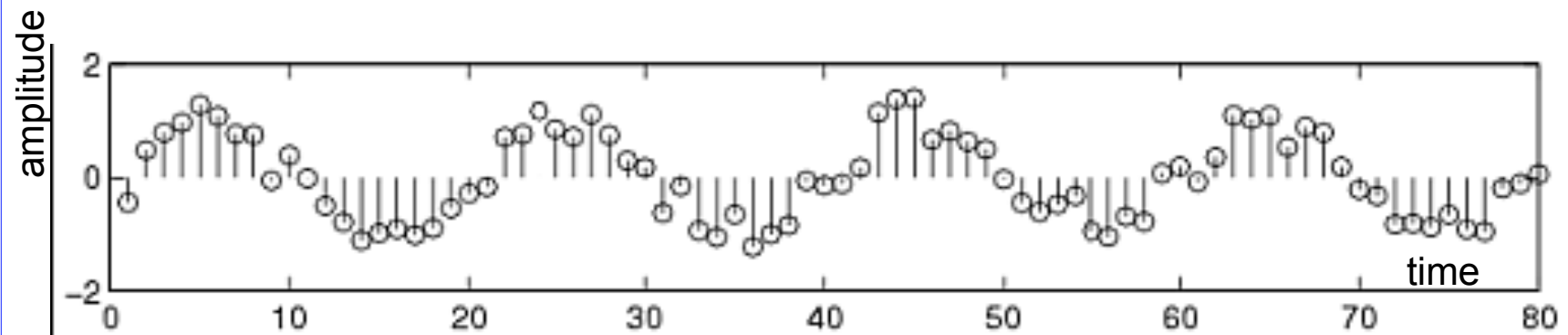
- Deterministic signal: a signal whose *physical description* is known completely
- A deterministic signal is a signal in which each value of the signal is fixed and can be determined by a mathematical expression, rule, or table.
- Because of this the future values of the signal can be calculated from past values with complete confidence.
  - There is *no uncertainty* about its amplitude values
  - Examples: signals defined through a mathematical function or graph
- Probabilistic (or random) signals: the amplitude values *cannot be predicted precisely* but are known only in terms of probabilistic descriptors
- The future values of a random signal cannot be accurately predicted and can usually only be guessed based on the averages of sets of signals
  - They are realization of a stochastic process for which a model could be available
  - Examples: EEG, evoked potentials, noise in CCD capture devices for digital cameras

# Example

- Deterministic signal



- Random signal



# Finite and Infinite length signals

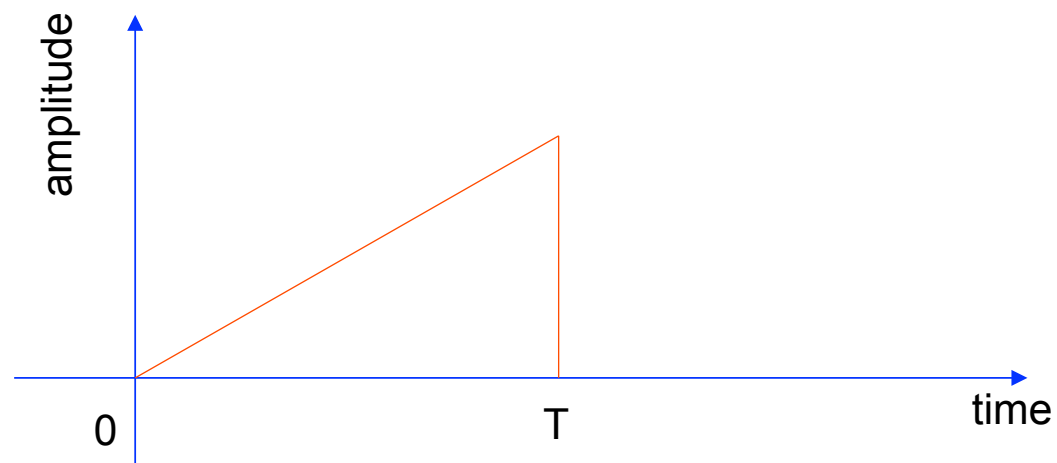
- A finite length signal is non-zero over a finite set of values of the independent variable

$$f = f(t), \forall t : t_1 \leq t \leq t_2$$
$$t_1 > -\infty, t_2 < +\infty$$

- An infinite length signal is non zero over an infinite set of values of the independent variable
  - For instance, a sinusoid  $f(t)=\sin(\omega t)$  is an infinite length signal

# Size of a signal: Norms

- "Size" indicates largeness or strength.
- We will use the mathematical concept of the norm to quantify this notion for both continuous-time and discrete-time signals.
- The energy is represented by the area under the curve (of the squared signal)



# Energy

- Signal energy

$$E_f = \int_{-\infty}^{+\infty} f^2(t) dt$$

$$E_f = \int_{-\infty}^{+\infty} |f(t)|^2 dt$$

- Generalized energy :  $L_p$  norm

- For  $p=2$  we get the energy ( $L_2$  norm)

$$\|f(t)\| = \left( \int (|f(t)|)^p dt \right)^{1/p}$$

$$1 \leq p < +\infty$$



# Power

- Power

- The power is the time average (mean) of the squared signal amplitude, that is the *mean-squared* value of  $f(t)$

$$P_f = \lim_{T \rightarrow \infty} \frac{1}{T} \int_{-T/2}^{+T/2} f^2(t) dt$$

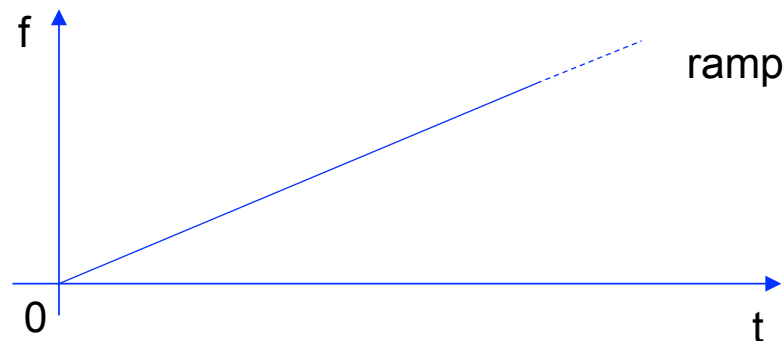
$$P_f = \lim_{T \rightarrow \infty} \frac{1}{T} \int_{-T/2}^{+T/2} |f(t)|^2 dt$$

# Power - Energy

- The square root of the power is the root mean square (*rms*) value
  - This is a very important quantity as it is the most widespread measure of similarity/dissimilarity among signals
  - It is the basis for the definition of the Signal to Noise Ratio (SNR)

$$SNR = 20 \log_{10} \left( \sqrt{\frac{P_{signal}}{P_{noise}}} \right)$$

- It is such that a constant signal whose amplitude is =rms holds the same power content of the signal itself
- There exists signals for which neither the energy nor the power are finite

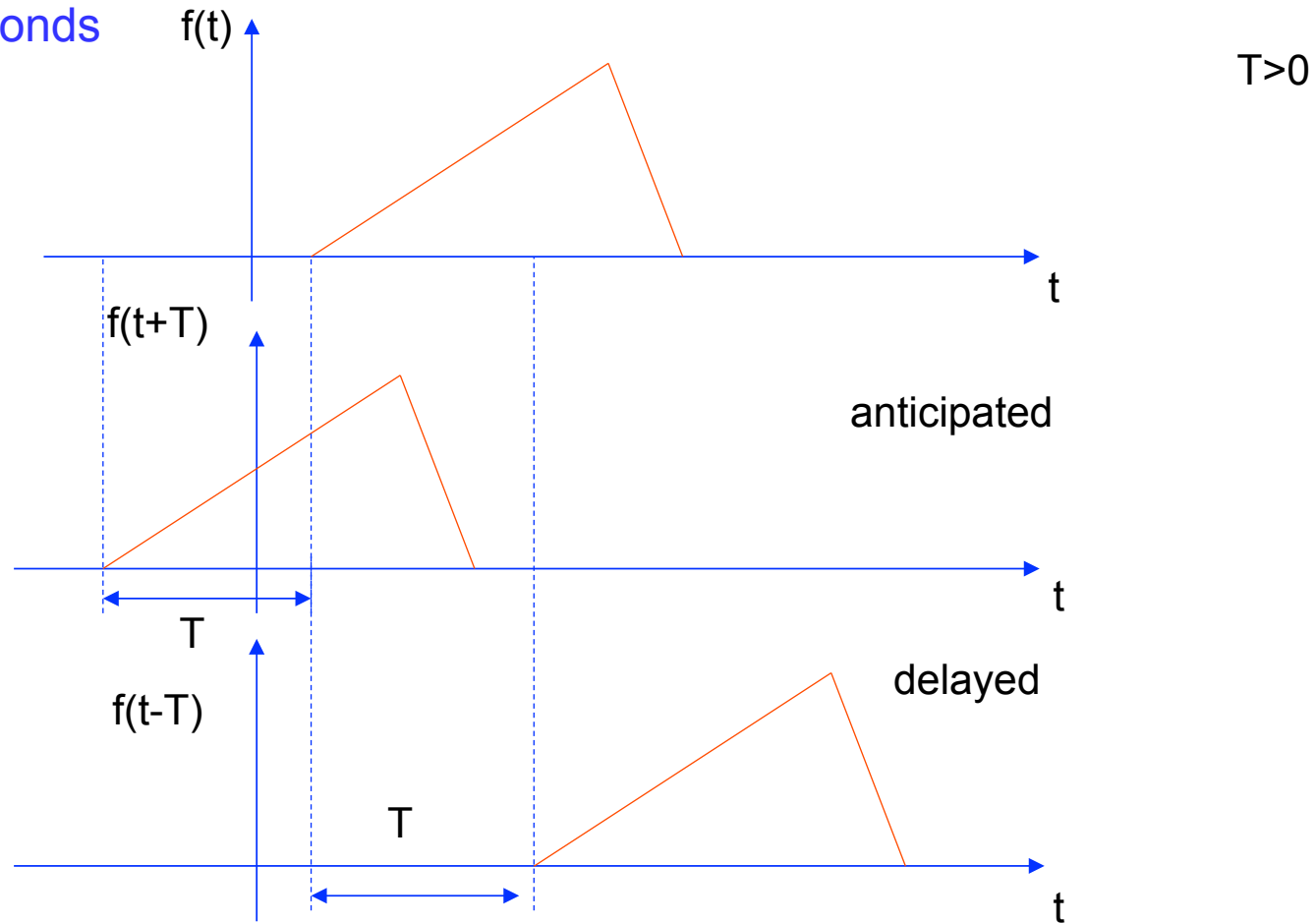


# Energy and Power signals

- A signal with finite energy is an energy signal
  - Necessary condition for a signal to be of energy type is that the amplitude goes to zero as the independent variable tends to infinity
- A signal with finite and different from zero power is a power signal
  - The mean of an entity averaged over an infinite interval exists if either the entity is periodic or it has some statistical regularity
  - A power signal has infinite energy and an energy signal has zero power
  - There exist signals that are neither power nor energy, such as the ramp
- All practical signals have finite energy and thus are energy signals
  - It is impossible to generate a real power signal because this would have infinite duration and infinite energy, which is not doable.

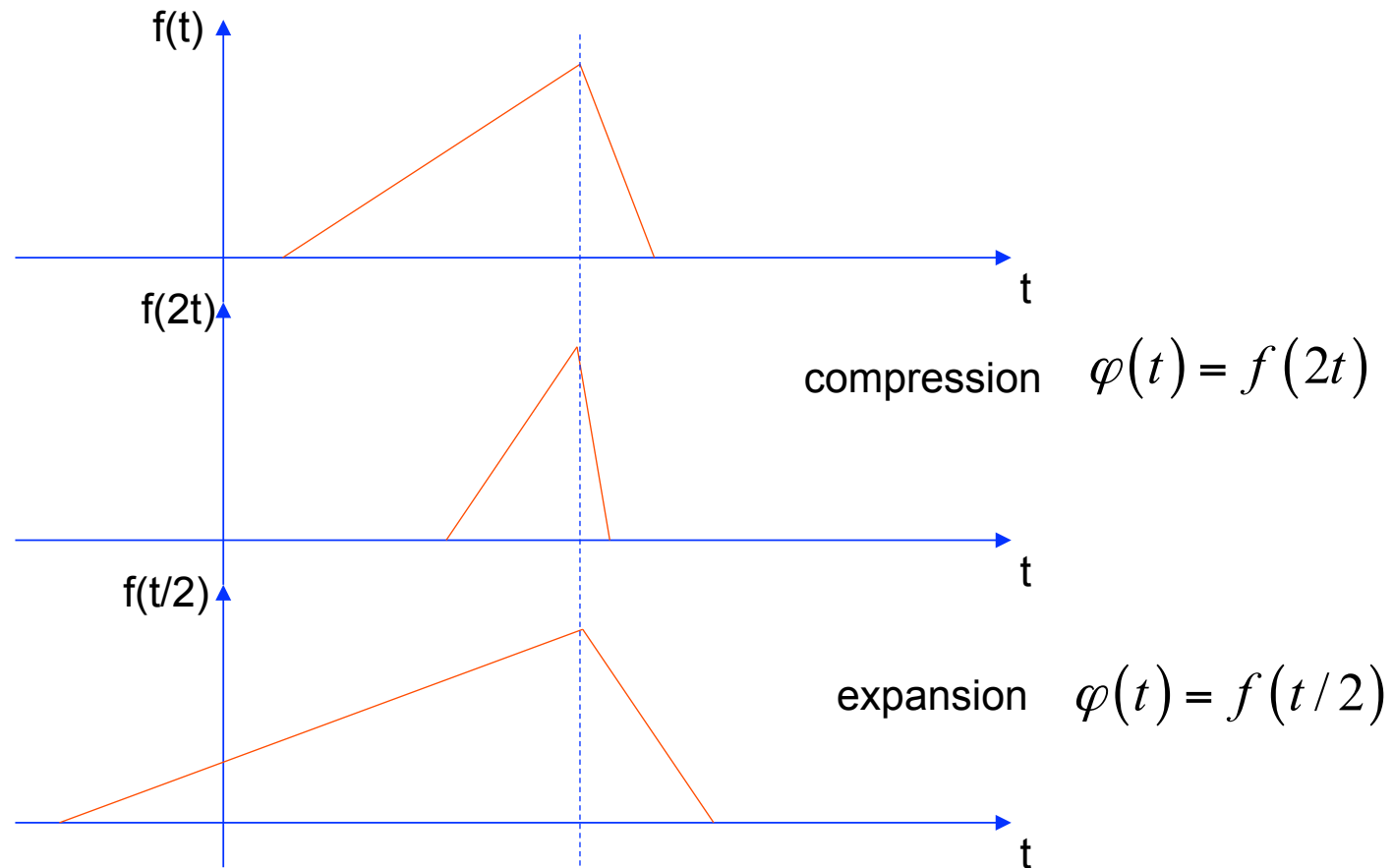
# Useful signal operations: shifting, scaling, inversion

- **Shifting:** consider a signal  $f(t)$  and the same signal delayed/anticipated by  $T$  seconds



# Useful signal operations: shifting, scaling, inversion

- (Time) Scaling: compression or expansion of a signal in time



# Useful signal operations: shifting, scaling, inversion

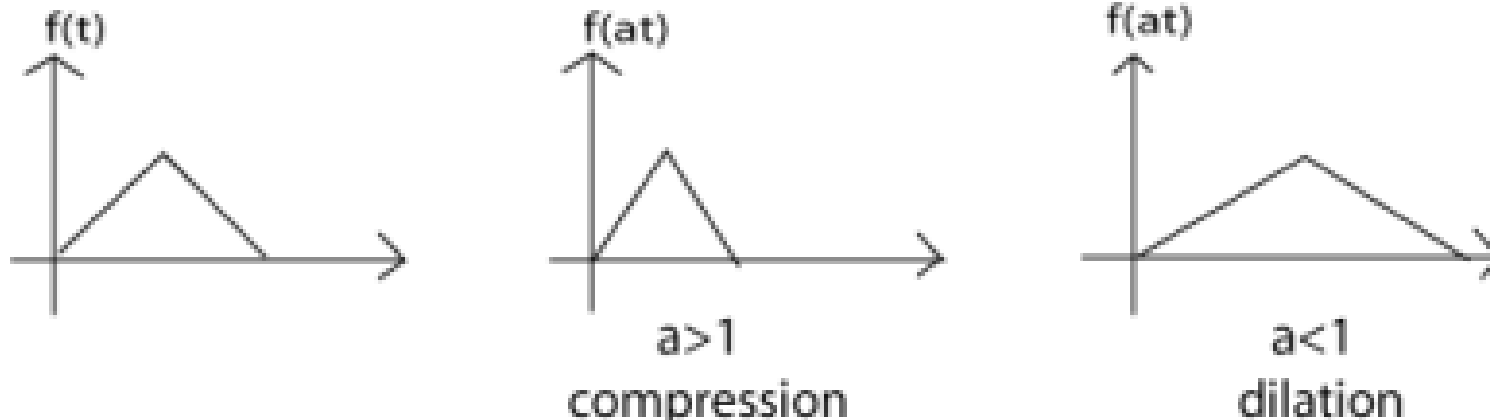
- Scaling: generalization

$$a > 1$$

$$\varphi(t) = f(at) \rightarrow \text{compressed version}$$

$$\varphi(t) = f\left(\frac{t}{a}\right) \rightarrow \text{dilated (or expanded) version}$$

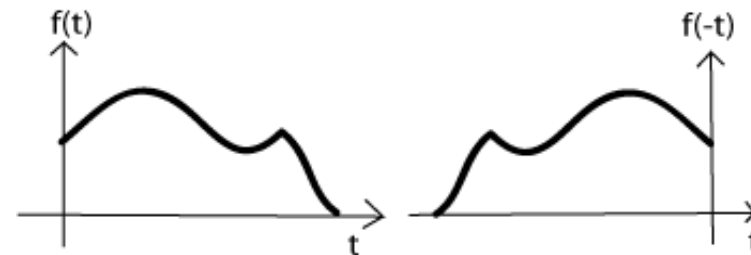
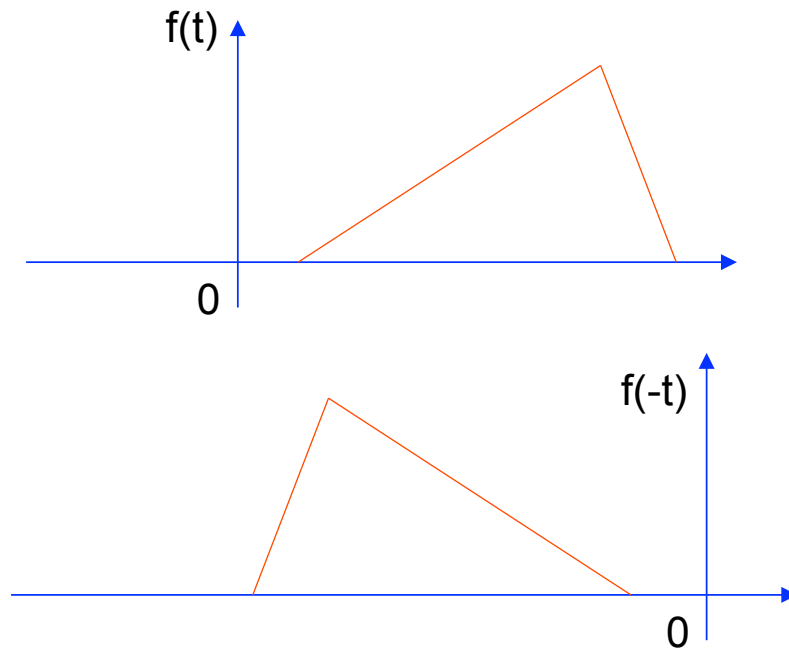
Viceversa for  $a < 1$



# Useful signal operations: shifting, scaling, inversion

- (Time) inversion: mirror image of  $f(t)$  about the vertical axis

$$\varphi(t) = f(-t)$$



# Useful signal operations: shifting, scaling, inversion

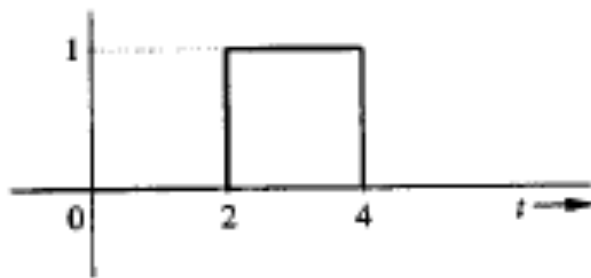
- Combined operations:  $f(t) \rightarrow f(at-b)$
- Two possible sequences of operations
  1. Time shift  $f(t)$  by  $t$  to obtain  $f(t-b)$ . Now time scale the shifted signal  $f(t-b)$  by  $a$  to obtain  $f(at-b)$ .
  2. Time scale  $f(t)$  by  $a$  to obtain  $f(at)$ . Now time shift  $f(at)$  by  $b/a$  to obtain  $f(at-b)$ .
    - Note that you have to replace  $t$  by  $(t-b/a)$  to obtain  $f(at-b)$  from  $f(at)$  when replacing  $t$  by the translated argument (namely  $t-b/a$ )



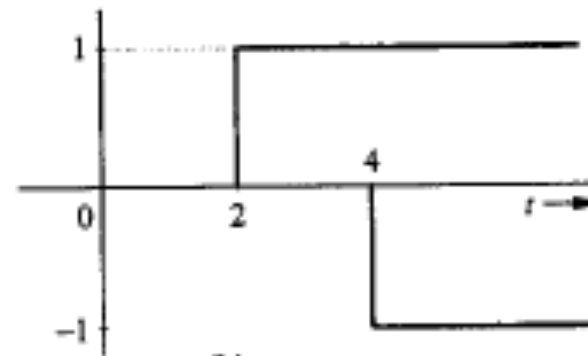
# Useful functions

- Unit step function
  - Useful for representing causal signals

$$u(t) = \begin{cases} 1 & t \geq 0 \\ 0 & t < 0 \end{cases}$$



(a)



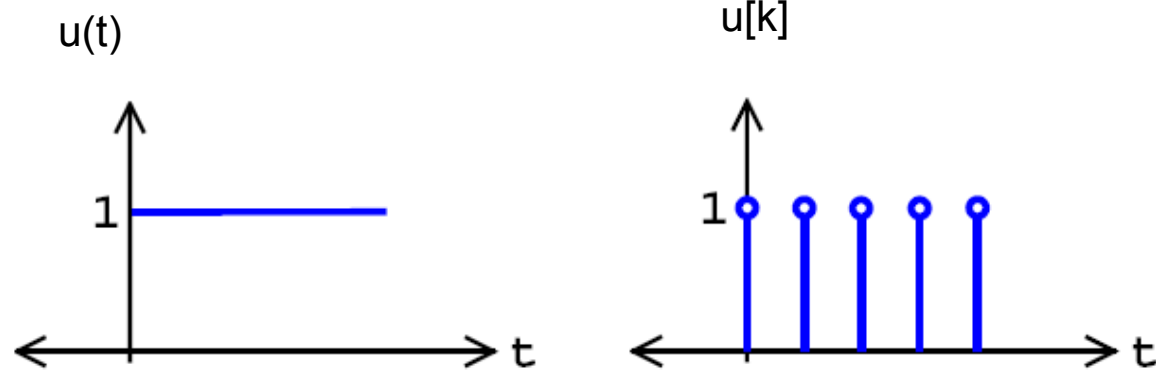
(b)

**Fig. 1.15** Representation of a rectangular pulse by step functions.

$$f(t) = u(t-2) - u(t-4)$$

# Useful functions

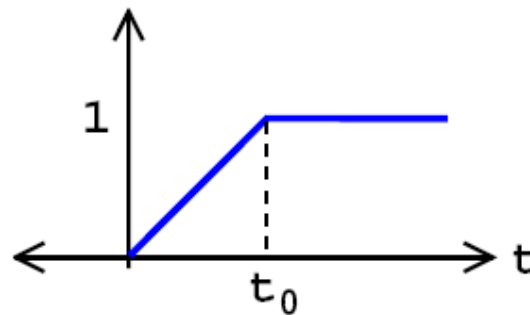
- Continuous and discrete time unit step functions



# Useful functions

- Ramp function (continuous time)

$$r(t) = \begin{cases} 0 & \text{if } t < 0 \\ \frac{t}{t_0} & \text{if } 0 \leq t \leq t_0 \\ 1 & \text{if } t > t_0 \end{cases}$$

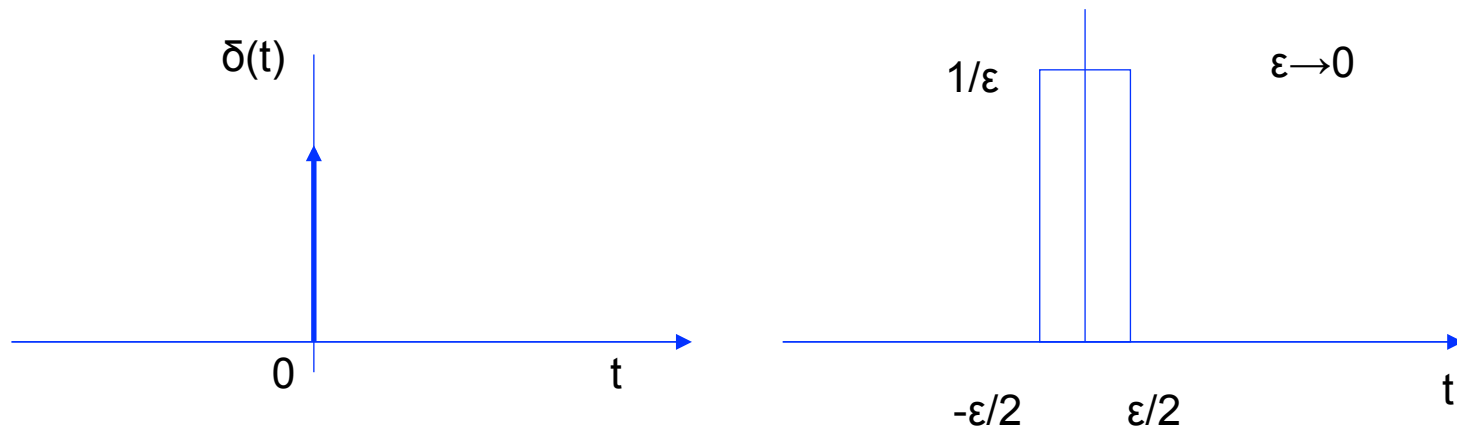


# Useful functions

- Unit impulse function

$$\delta(t) = 0 \quad t \neq 0$$

$$\int_{-\infty}^{+\infty} \delta(t) dt = 1$$



# Properties of the unit impulse function

- Multiplication of a function by impulse

$$\phi(t)\delta(t) = \phi(0)\delta(t)$$

$$\phi(t)\delta(t-T) = \phi(T)\delta(t-T)$$

- Sampling property of the unit function

$$\int_{-\infty}^{+\infty} \phi(t)\delta(t)dt = \int_{-\infty}^{+\infty} \phi(0)\delta(t)dt = \phi(0) \int_{-\infty}^{+\infty} \delta(t)dt = \phi(0)$$

$$\int_{-\infty}^{+\infty} \phi(t)\delta(t-T)dt = \phi(T)$$

- The area under the curve obtained by the product of the unit impulse function shifted by T and  $\phi(t)$  is the value of the function  $\phi(t)$  for  $t=T$

# Properties of the unit impulse function

- The unit step function is the integral of the unit impulse function

$$\frac{du}{dt} = \delta(t)$$

$$\int_{-\infty}^t \delta(t) dt = u(t)$$

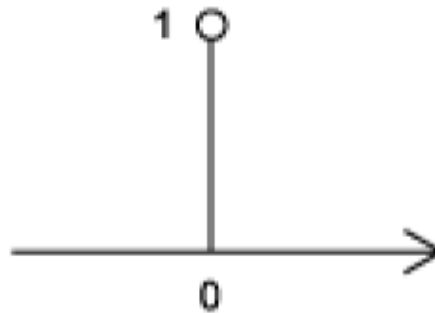
– Thus

$$\int_{-\infty}^t \delta(t) dt = u(t) = \begin{cases} 0 & t < 0 \\ 1 & t \geq 0 \end{cases}$$

# Properties of the unit impulse function

- Discrete time impulse function

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



# Useful functions

- Continuous time complex exponential

$$f(t) = Ae^{j\omega t}$$

- Euler's relations

$$Ae^{j\omega t} = A\cos(\omega t) + j(A\sin(\omega t))$$

$$\cos(\omega t) = \frac{e^{j\omega t} + e^{-j\omega t}}{2}$$

$$\sin(\omega t) = \frac{e^{j\omega t} - e^{-j\omega t}}{2j}$$

$$e^{j\omega t} = \cos(\omega t) + j\sin(\omega t)$$

- Discrete time complex exponential

–  $k=nT$

$$\begin{aligned} f[n] &= Be^{snT} \\ &= Be^{j\omega nT} \end{aligned}$$



## Useful functions

- Exponential function  $e^{st}$ 
  - Generalization of the function  $e^{j\omega t}$

$$s = \sigma + j\omega$$

Therefore

$$e^{st} = e^{(\sigma+j\omega)t} = e^{\sigma t} e^{j\omega t} = e^{\sigma t} (\cos \omega t + j \sin \omega t) \quad (1.30a)$$

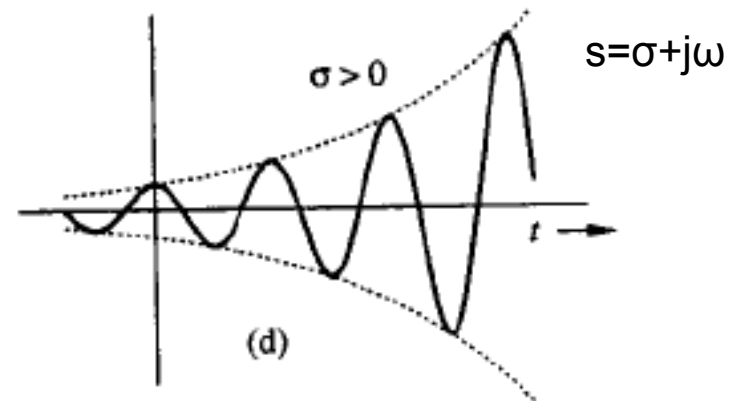
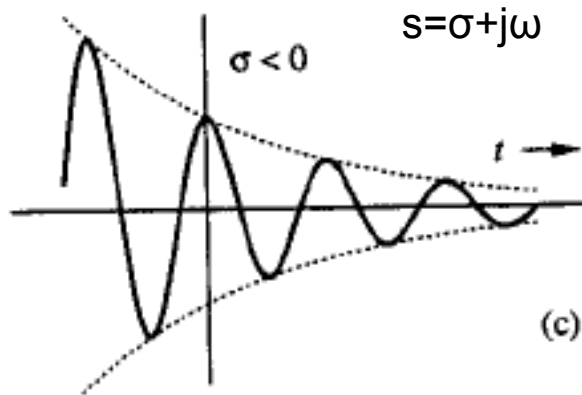
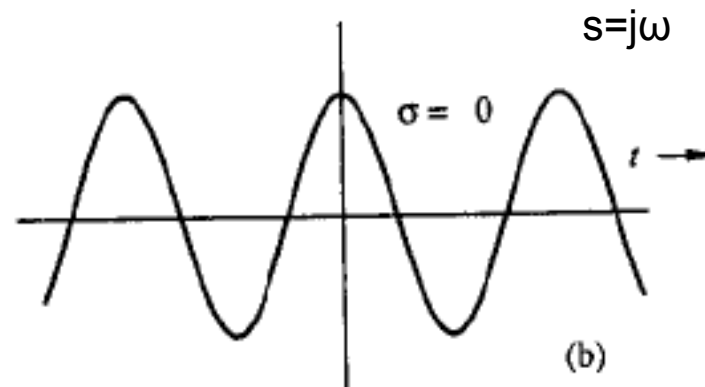
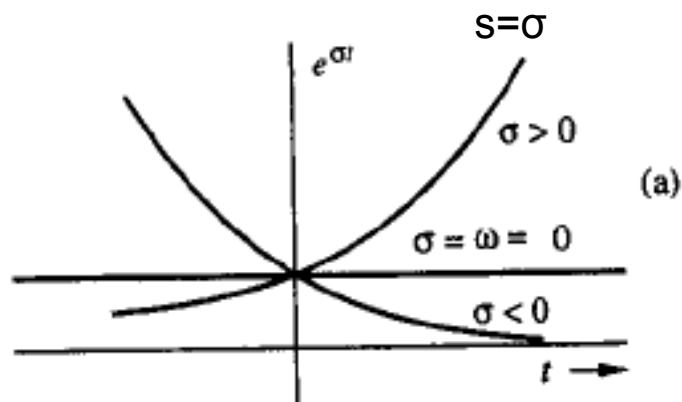
If  $s^* = \sigma - j\omega$  (the conjugate of  $s$ ), then

$$e^{s^*t} = e^{\sigma-j\omega t} = e^{\sigma t} e^{-j\omega t} = e^{\sigma t} (\cos \omega t - j \sin \omega t) \quad (1.30b)$$

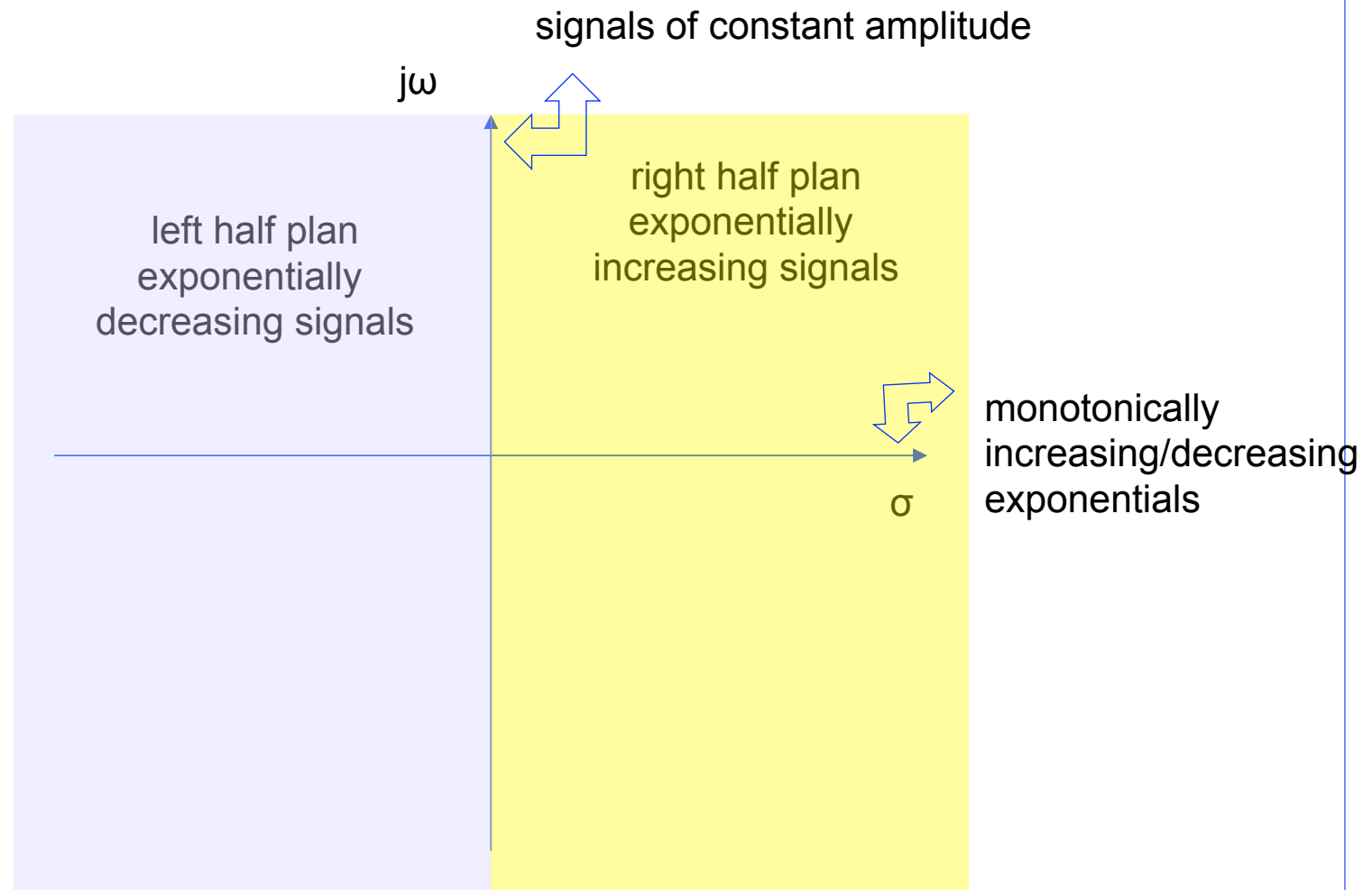
and

$$e^{\sigma t} \cos \omega t = \frac{1}{2}(e^{st} + e^{s^*t}) \quad (1.30c)$$

# The exponential function



# Complex frequency plan

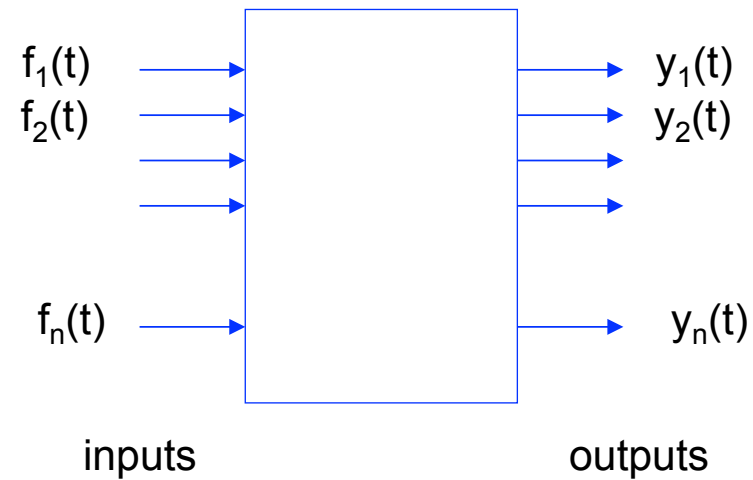


# Basics of Linear Systems

## 2D Linear Systems

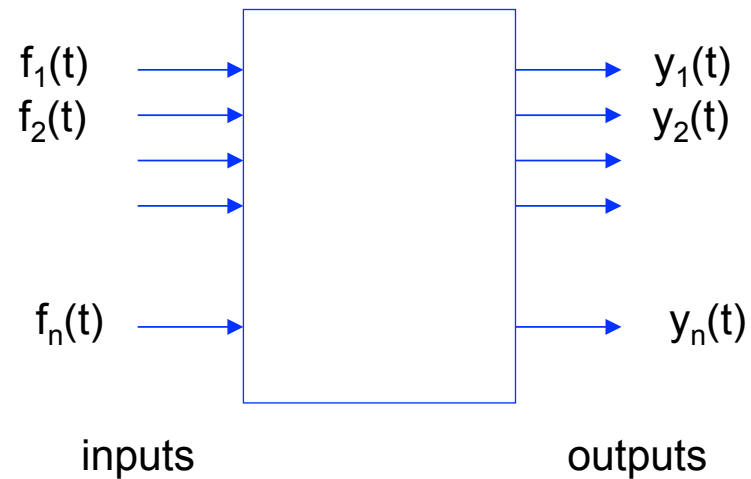
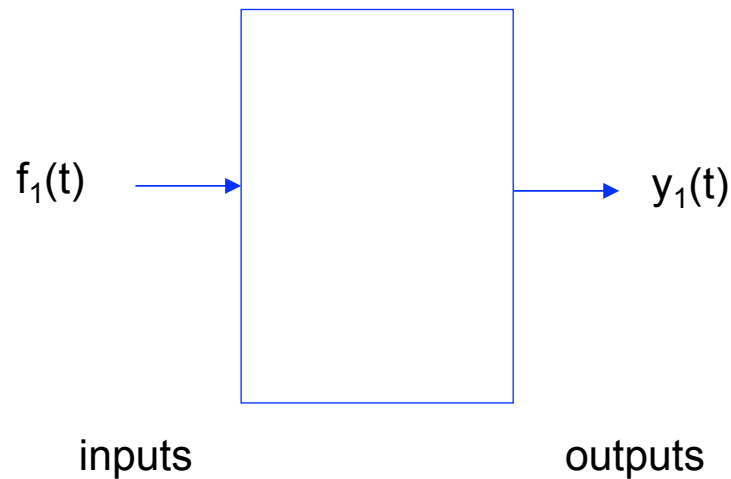
# Systems

- A system is characterized by
  - inputs
  - outputs
  - rules of operation (mathematical model of the system)



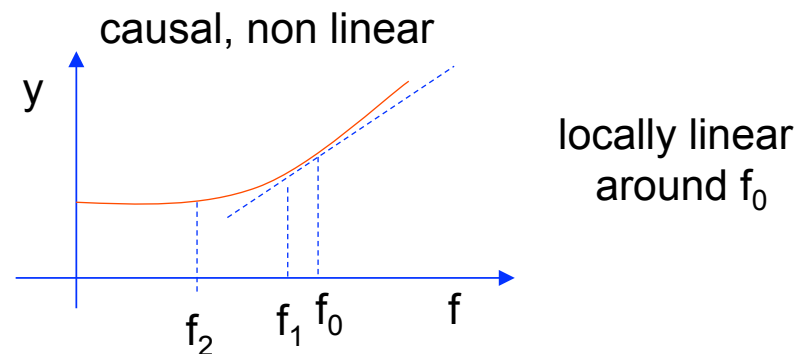
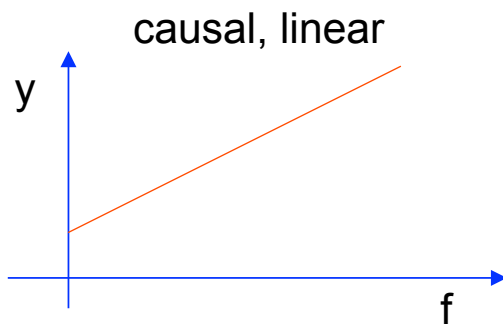
# Systems

- Study of systems: mathematical modeling, analysis, design
  - Analysis: how to determine the system output given the input and the system mathematical model
  - design or synthesis: how to design a system that will produce the desired set of outputs for given inputs
- SISO: single input single output - MIMO: multiple input multiple output



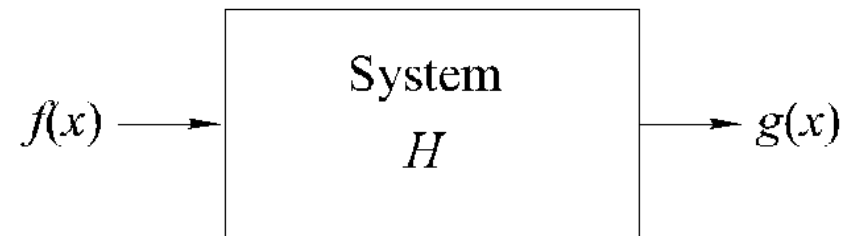
# Response of a linear system

- Total response = Zero-input response + Zero-state response
  - The output of a system for  $t \geq 0$  is the result of two independent causes: the initial conditions of the system (or system state) at  $t=0$  and the input  $f(t)$  for  $t \geq 0$ .
  - Because of linearity, the total response is the sum of the responses due to those two causes
  - The zero-input response is only due to the initial conditions and the zero-state response is only due to the input signal
  - This is called decomposition property
- Real systems are *locally* linear
  - Respond linearly to small signals and non-linearly to large signals



# Review: Linear Systems

- We define a system as a unit that converts an input function into an output function

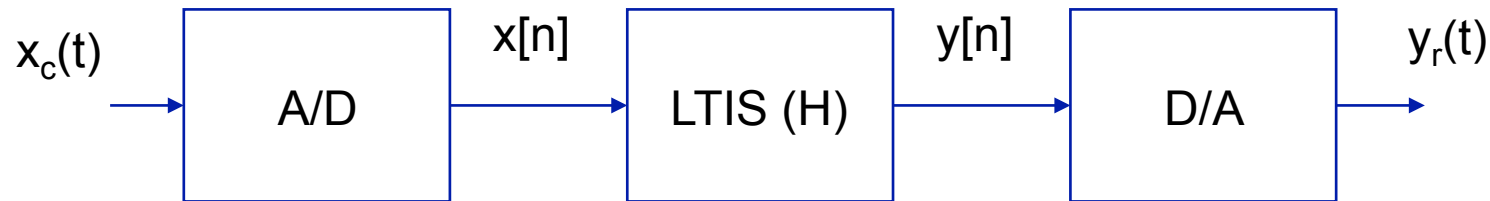


$$g(x) = H[f(x)]$$

Independent System operator or Transfer function variable



# Linear Time Invariant Discrete Time Systems



$$Y(e^{j\omega}) = H(e^{j\omega})X(e^{j\omega})$$

$$Y_r(j\Omega) = H(j\Omega)X_c(j\Omega)$$

$$H(j\Omega) = \begin{cases} H(j\Omega) & |\Omega| < \pi/T \\ 0 & |\Omega| \geq \pi/T \end{cases}$$

IF

- The input signal is bandlimited
- The Nyquist condition for sampling is met
- The digital system is linear and time invariant

THEN

The overall continuous time system is equivalent to a LTIS whose frequency response is H.

# Overview of Linear Systems

- Let

$$g_i(x) = H[f_i(x)]$$

where  $f_i(x)$  is an arbitrary input in the class of all inputs  $\{f(x)\}$ , and  $g_i(x)$  is the corresponding output.

- If

$$\begin{aligned} H[a_i f_i(x) + a_j f_j(x)] &= a_i H[f_i(x)] + a_j H[f_j(x)] \\ &= a_i g_i(x) + a_j g_j(x) \end{aligned}$$

Then the system  $H$  is called a *linear system*.

- A linear system has the properties of *additivity* and *homogeneity*.

# Linear Systems

- The system  $H$  is called *shift invariant* if

$$g_i(x) = H[f_i(x)] \text{ implies that } g_i(x + x_0) = H[f_i(x + x_0)]$$

for all  $f_i(x) \in \{f(x)\}$  and for all  $x_0$ .

- This means that offsetting the independent variable of the input by  $x_0$  causes the same offset in the independent variable of the output. Hence, the input-output relationship remains the same.

# Linear Systems

- The operator  $H$  is said to be *causal*, and hence the system described by  $H$  is a *causal system*, if there is no output before there is an input. In other words,

$$f(x) = 0 \text{ for } x < x_0 \text{ implies that } g(x) = H[f(x)] = 0 \text{ for } x < x_0.$$

- A linear system  $H$  is said to be *stable* if its response to any bounded input is bounded. That is, if

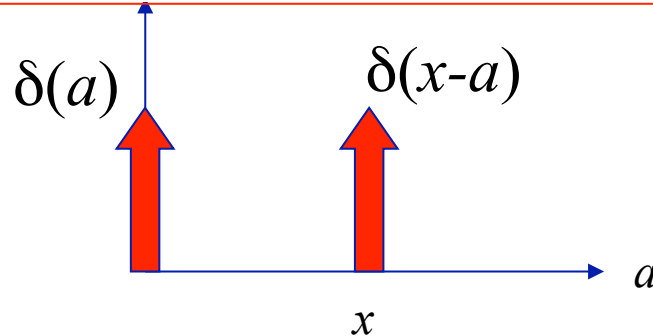
$$|f(x)| < K \text{ implies that } |g(x)| < cK$$

where  $K$  and  $c$  are constants.

# Linear Systems

- A *unit impulse function*, denoted  $\delta(a)$ , is *defined* by the expression

$$\int_{-\infty}^{\infty} f(a)\delta(x-a)da = f(x).$$



- The response of a system to a unit impulse function is called the *impulse response* of the system

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

# Linear Systems

- If  $H$  is a linear shift-invariant system, then we can find its response to any input signal  $f(x)$  as follows:

$$g(x) = \int_{-\infty}^{\infty} f(\alpha)h(x - \alpha)d\alpha.$$

70

- This expression is called the *convolution integral*. It states that the response of a linear, fixed-parameter system is completely characterized by the convolution of the input with the system impulse response.

# Linear Systems

- Convolution of two functions of a continuous variable is defined as

$$f(x) * h(x) = \int_{-\infty}^{\infty} f(\alpha)h(x - \alpha)d\alpha$$

- In the discrete case

$$f[n] * h[n] = \sum_{m=-\infty}^{\infty} f[m]h[n - m]$$

# Linear Systems

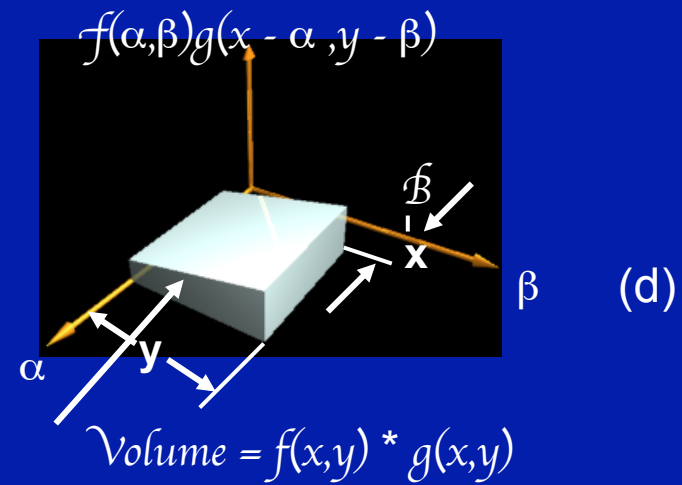
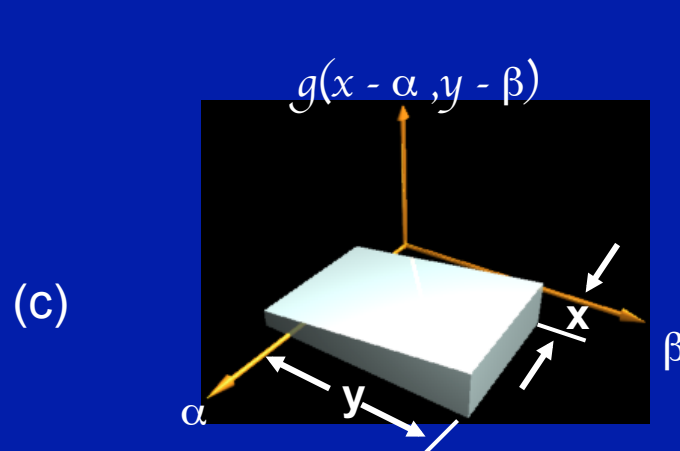
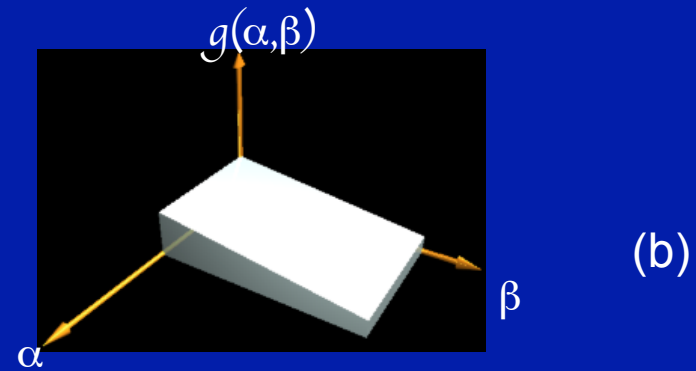
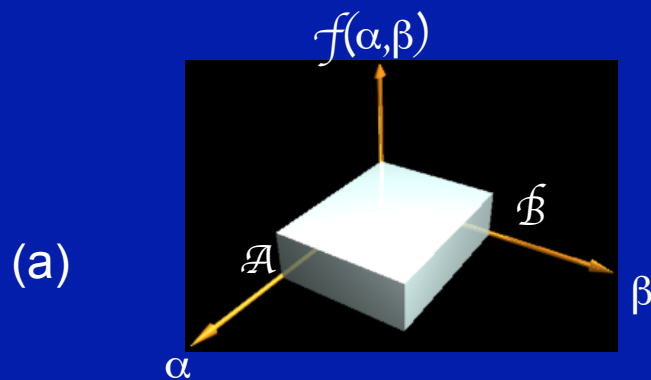
- In the 2D discrete case

$$f[n_1, n_2] * h[n_1, n_2] = \sum_{m_1=-\infty}^{\infty} \sum_{m_2=-\infty}^{\infty} f[m_1, m_2] h[n_1 - m_1, n_2 - m_2]$$

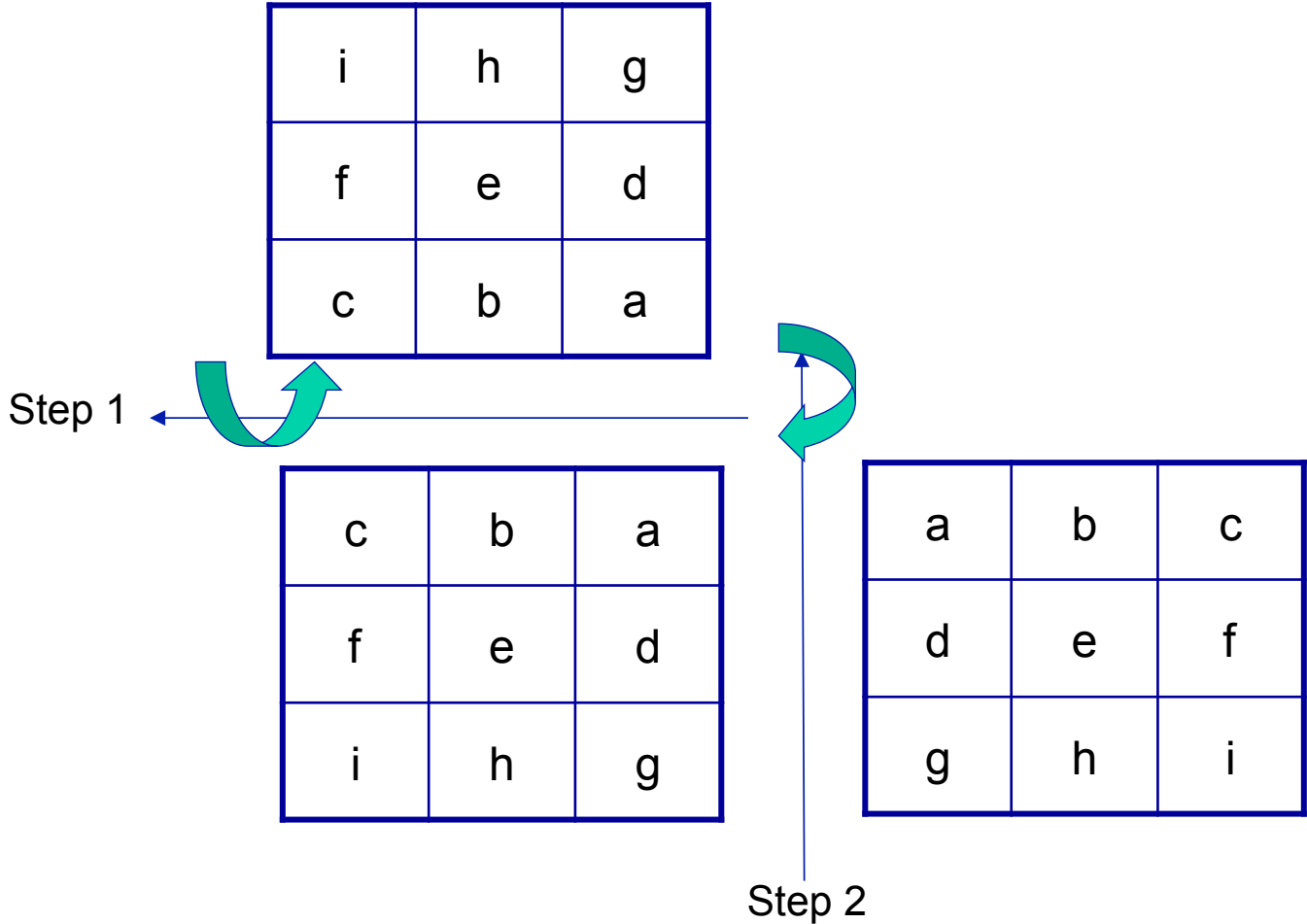
$h[n_1, n_2]$  is a linear filter.



# Illustration of the folding, displacement, and multiplication steps needed to perform two-dimensional convolution



# Matrix perspective



# Convolution Example

h

1	-1	-1
1	2	-1
1	1	1

Rotate

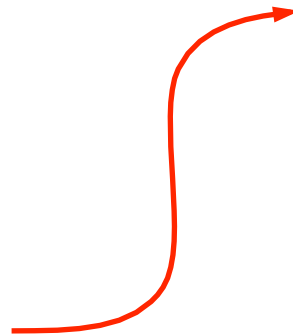


1	1	1
-1	2	1
-1	-1	1

$$f[n_1, n_2] ** h[n_1, n_2] = \sum_{m_1=-\infty}^{\infty} \sum_{m_2=-\infty}^{\infty} f[m_1, m_2] h[n_1 - m_1, n_2 - m_2]$$

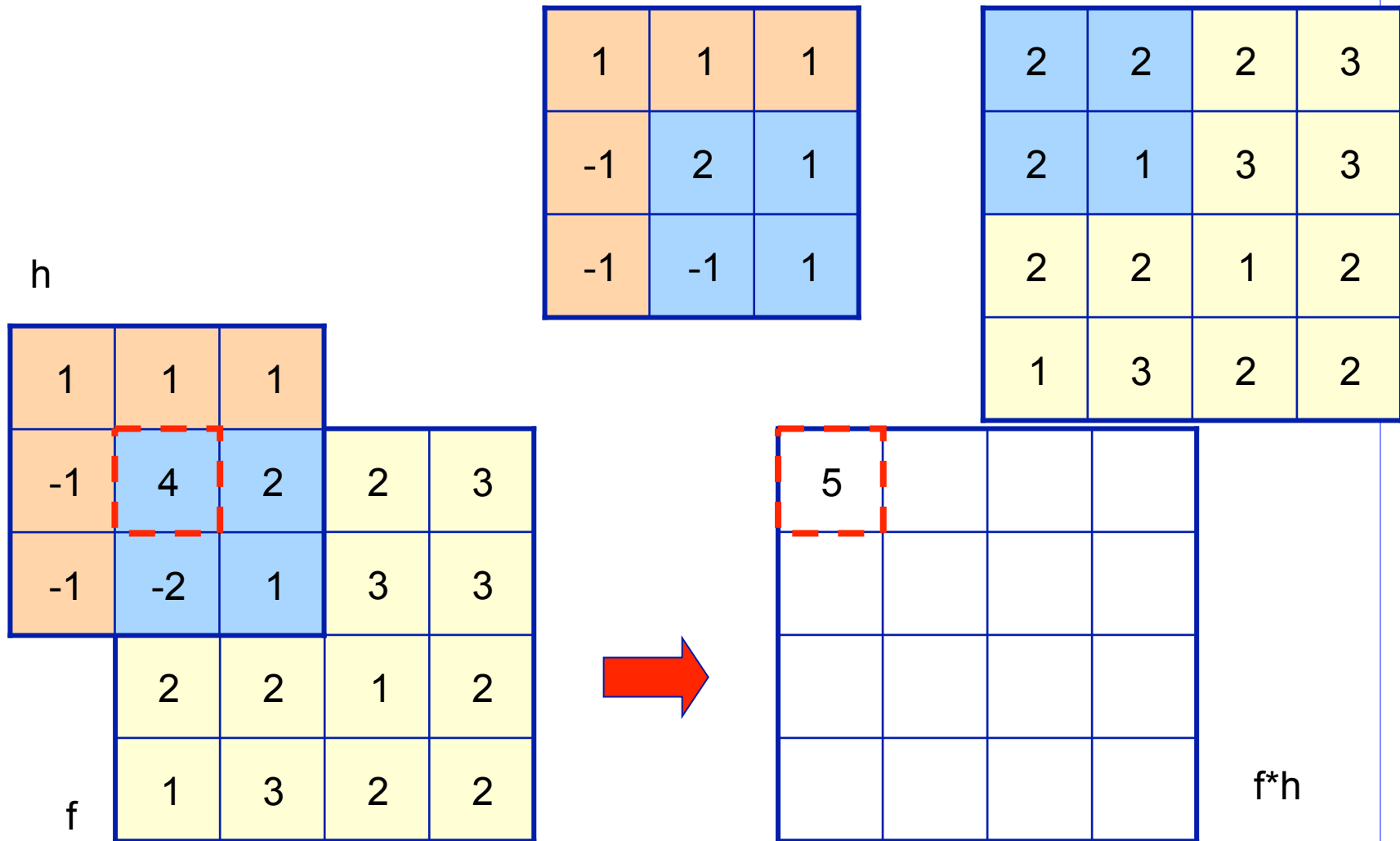
f

2	2	2	3
2	1	3	3
2	2	1	2
1	3	2	2



From C. Rasmussen, U. of Delaware

# Convolution Example



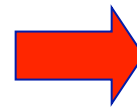
# Convolution Example

h

1	1	1
-1	2	1
-1	-1	1

2	2	2	3
2	1	3	3
2	2	1	2
1	3	2	2

1	1	1	
-2	4	2	3
-2	-1	3	3
2	2	1	2
1	3	2	2



5	4		

f\*h

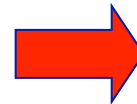
# Convolution Example

1	1	1
-1	2	1
-1	-1	1

2	2	2	3
2	1	3	3
2	2	1	2
1	3	2	2

h

	1	1	1
2	-2	4	3
2	-1	-3	3
2	2	1	2
1	3	2	2



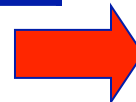
5	4	4	

# Convolution Example

1	1	1
-1	2	1
-1	-1	1

2	2	2	3
2	1	3	3
2	2	1	2
1	3	2	2

		1	1	1
2	2	-2	6	1
2	1	-3	-3	1
2	2	1	2	
1	3	2	2	



5	4	4	-2

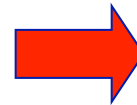
# Convolution Example

1	1	1
-1	2	1
-1	-1	1

2	2	2	3
2	1	3	3
2	2	1	2
1	3	2	2

h

1	2	2	2	3
-1	4	1	3	3
-1	-2	2	1	2
	1	3	2	2



5	4	4	-2
9			

80

f

f\*h



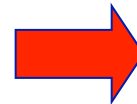
# Convolution Example

1	1	1
-1	2	1
-1	-1	1

2	2	2	3
2	1	3	3
2	2	1	2
1	3	2	2

h

2	2	2	3
-2	2	3	3
-2	-2	1	2
1	3	2	2



5	4	4	-2
9	6		