

**MAC317**

# **Introdução ao Processamento de Sinais Digitais**

---

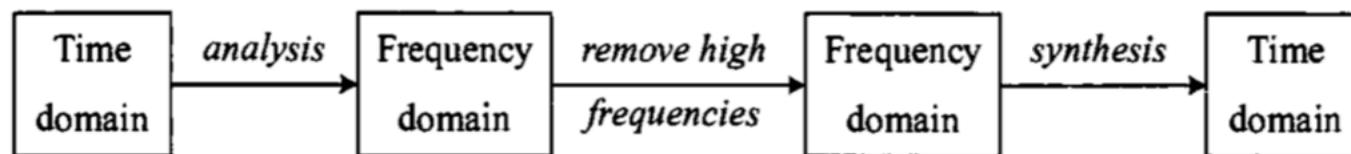
**Prof. Marcel P. Jackowski**

`mjack@ime.usp.br`

**Aula #5: Transformada discreta de Fourier**

# Tempo/espacô vs. frequênciâ

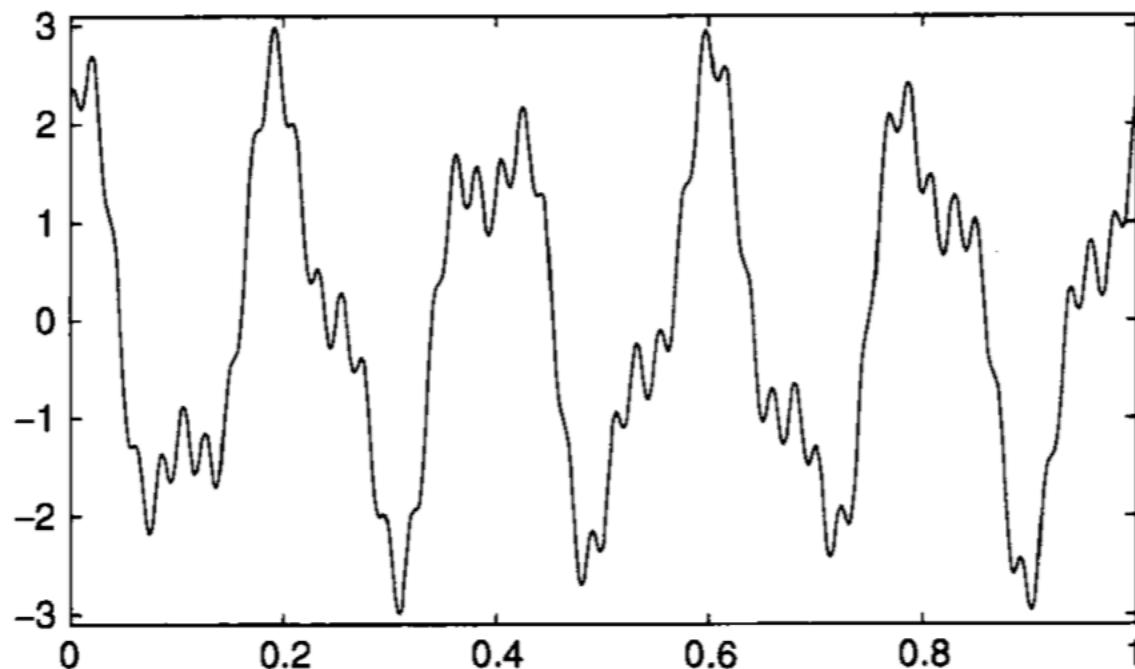
- Many operations are easier, both computationally and conceptually, in the frequency domain.
- We transform the time domain signal to the frequency domain, perform the operation of interest there, and then transform the altered signal back to the time domain.



**FIGURE 2.1** Processing a signal in the frequency domain.

# Exemplo

---



$$x(t) = 2.0 \cos(2\pi \cdot 5t) + 0.8 \sin(2\pi \cdot 12t) + 0.3 \cos(2\pi \cdot 47t)$$

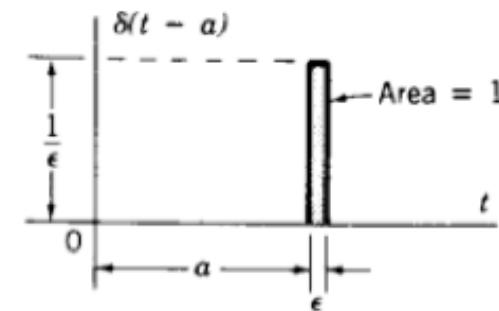
# Função impulso

---

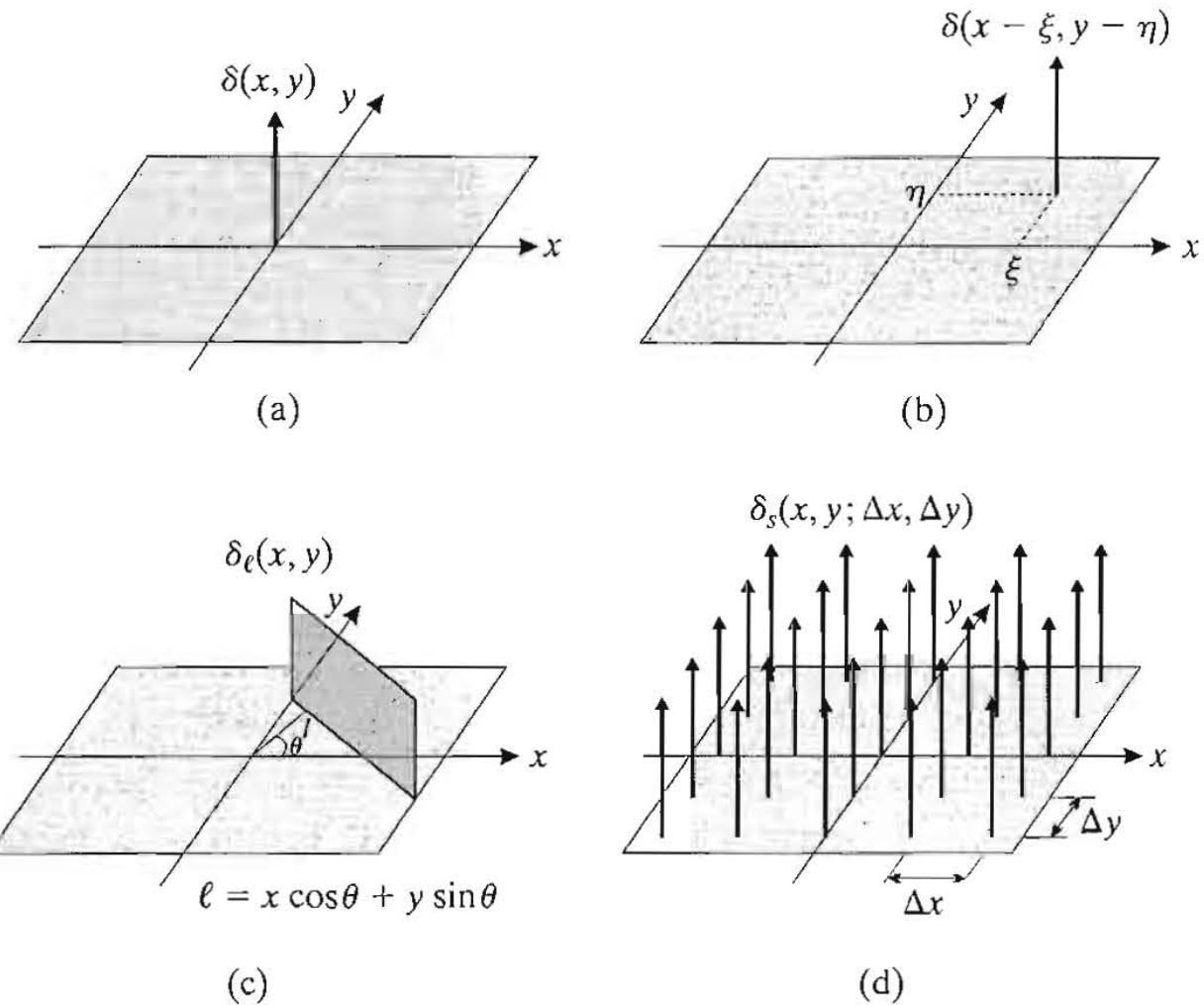
- Também conhecida como função Dirac ou função delta
- Modela a propriedade de uma fonte puntual tendo largura infinitesimal e área (volume) unitária

$$\delta(x - x_0) = 0 \quad \text{for } x \neq x_0,$$

$$\int_{-\infty}^{+\infty} \delta(x - x_0) \, dx = 1,$$



# Função impulso em 2D



**Figure 2.3**

Signals derived from the point impulse: (a) point impulse  $\delta(x, y)$ , (b) shifted point impulse  $\delta(x - \xi, y - \eta)$ , (c) line impulse  $\delta_\ell(x, y)$ , and (d) sampling function  $\delta_s(x, y; \Delta x, \Delta y)$ .

# Filtro “peneira”

---

*sifting* let  $s(x)$  be continuous at  $x = x_0$ , then

$$\int_{-\infty}^{+\infty} s(x) \delta(x - x_0) dx = s(x_0);$$

*scaling*

$$\int_{-\infty}^{+\infty} A \delta(x) dx = A,$$

this is a special case of sifting.

$$\iint_{-\infty}^{+\infty} s(\vec{r}) \delta(\vec{r} - \vec{r}_0) d\vec{r} = s(\vec{r}_0).$$

# Sinais importantes

---

Exponential  $\exp$

$$\exp(ax) = e^{ax}$$

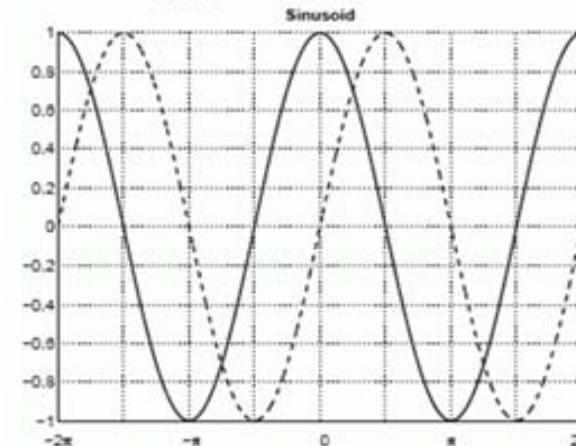
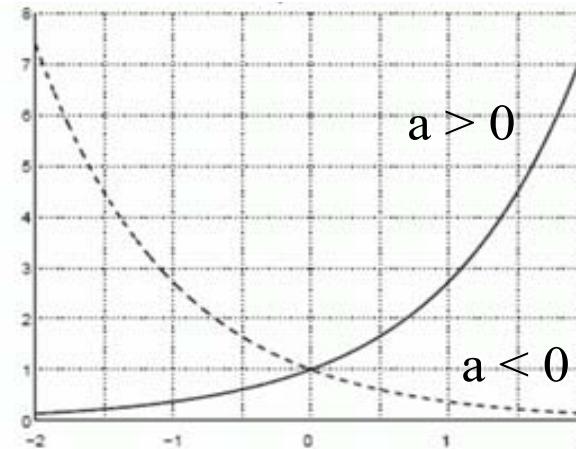
- when  $a > 0$  then  $\exp$  increases with increasing  $x$
- when  $a < 0$  then  $\exp$  approximates 0 with increasing  $x$

Complex exponential / sinusoid:

$$Ae^{i(2\pi kx + \phi)} = A(\cos(2\pi kx + \phi) + i \sin(2\pi kx + \phi))$$

As before

- the  $\cos$  term is the signal's real part
- the  $\sin$  term is the signal's imaginary part
- $A$  is the amplitude,  $\phi$  the phase shift,  $k$  determines the frequency



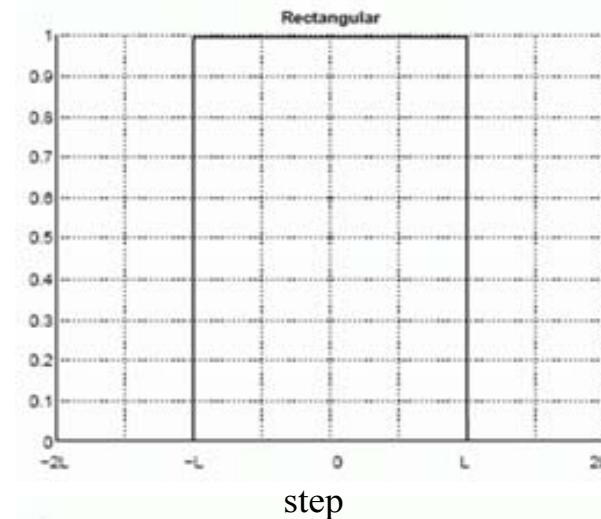
# Funções adicionais

---

Rectangular function:

$$\Pi\left(\frac{x}{2L}\right) = 1 \quad \text{for } |x| < L$$

$$\begin{aligned} &= \frac{1}{2} \quad \text{for } |x| = L \\ &= 0 \quad \text{for } |x| > L \end{aligned}$$

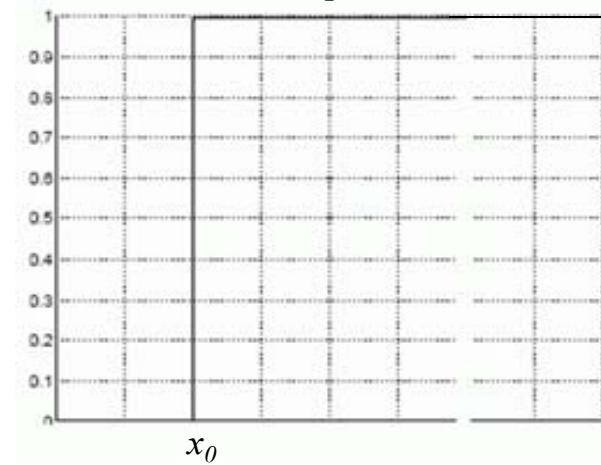


Step function:

$$u(x - x_0) = 0 \quad \text{for } x < x_0$$

$$= \frac{1}{2} \quad \text{for } x = x_0$$

$$= 1 \quad \text{for } x > x_0$$



# Funções adicionais

---

Triangular function:

$$\begin{aligned} Tri\left(\frac{x}{2L}\right) &= 1 - \frac{|x|}{L} && \text{for } |x| < L \\ &= 0 && \text{for } |x| > L \end{aligned}$$

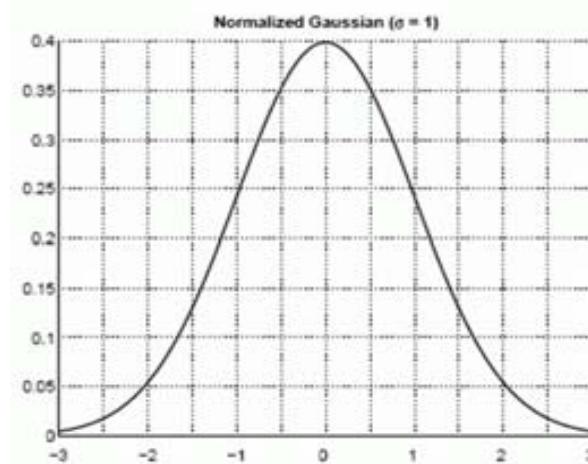
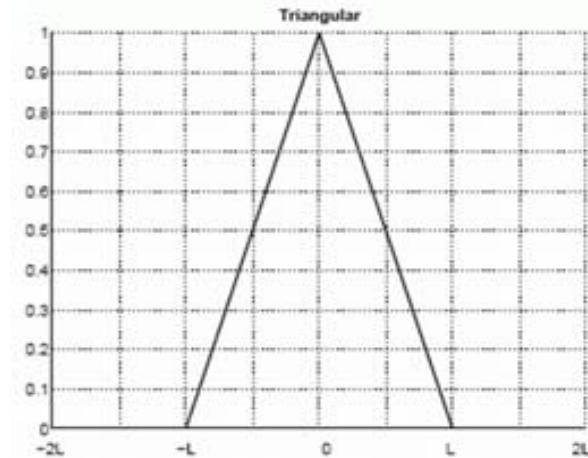
Normalized Gaussian:

$$G_n(x) = \frac{1}{\sqrt{2\pi}\sigma} e^{\frac{-(x-\mu)^2}{2\sigma^2}}$$

$\mu$  is the mean

$\sigma$  is the standard deviation

- normalized means that the integral for all  $x$  is 1



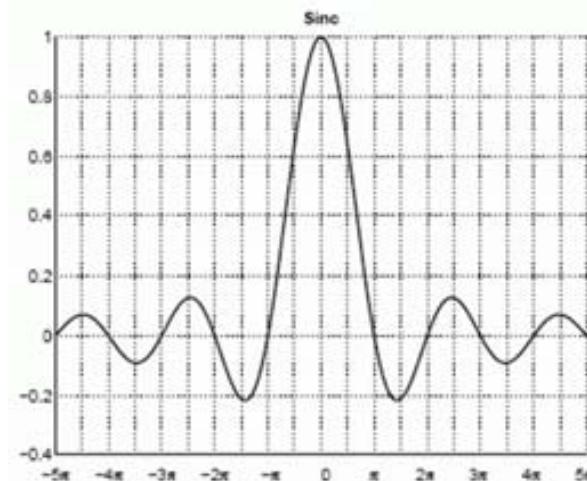
# Funções adicionais

---

*Sinc* function:

$$\text{sinc}(x) = \frac{\sin(\pi x)}{\pi x}$$

- $\text{sinc}(0) = 1$  (L'Hopital's rule)



# Separabilidade

---

- Um sinal  $f(x,y)$  é separável se existirem dois sinais unidimensionais  $f_1(x)$  e  $f_2(y)$  tais que:
  - $f(x,y) = f_1(x) f_2(y)$
- São limitados, pois podem somente modelar variações independentes em cada variável
- São apropriados em certas situações, pois são mais simples de operar do que funções em 2D ou 3D.

# Convolução

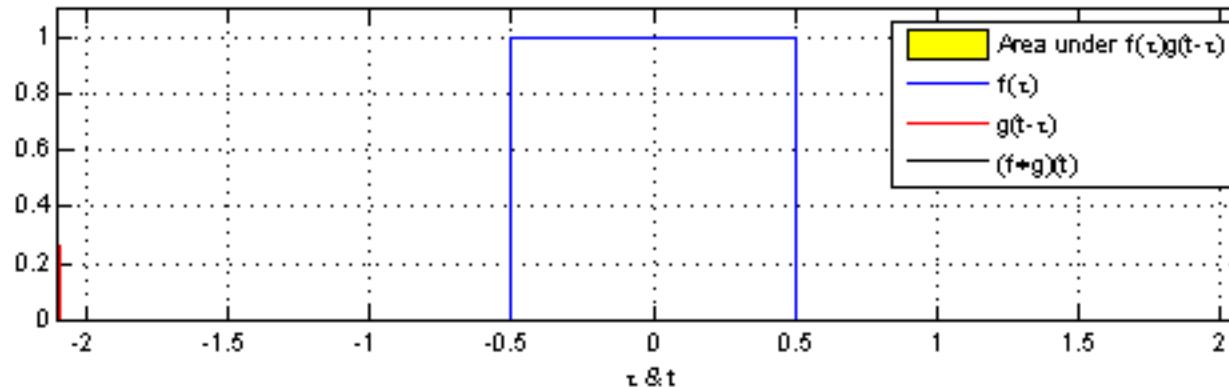
---

The expression

$$s_o(x) = \int_{-\infty}^{+\infty} s_i(\xi)h(x - \xi)d\xi = s_i * h$$

is called *convolution*, defined as:

$$s_1(x) * s_2(x) = \int_{-\infty}^{+\infty} s_1(\xi)s_2(x - \xi)d\xi = \int_{-\infty}^{+\infty} s_1(x - \xi)s_2(\xi)d\xi$$



# Sinais 2-D

---

$$s_1(x, y) * s_2(x, y)$$

$$= \iint_{-\infty}^{+\infty} s_1(x - \xi, y - \zeta) s_2(\xi, \zeta) d\xi d\zeta,$$

- Propriedades:

- $s_1 * s_2 = s_2 * s_1$  (comutativa)
- $(s_1 * s_2) * s_3 = (s_1 * s_2) * s_3$  (associativa)
- $s_1 * (s_2 + s_3) = s_1 * s_2 + s_1 * s_3$  (distributiva)

# Resposta de sistema

---

Now assume the input is a complex sinusoid with  $Ae^{2\pi i kx}$  then:

$$\begin{aligned} s_0(x) &= \int_{-\infty}^{+\infty} Ae^{2\pi i k(x-\xi)} h(\xi) d\xi \\ &= Ae^{2\pi i kx} \int_{-\infty}^{+\infty} e^{-2\pi i k\xi} h(\xi) d\xi \\ &= Ae^{2\pi i kx} H \end{aligned}$$

for now, assume  $\varphi=0$

$H$  is called the *Fourier Transform* of  $h(x)$ :

$$H = \int_{-\infty}^{+\infty} e^{-2\pi i k\xi} h(\xi) d\xi$$

- $H$  is also often called the *transfer function* or *filter*

# Observações

---

$H$  scales, and maybe phase-shifts, the input sinusoid  $S_i$

In essence, we have now two alternative representations:

- determine the effect of  $L$  on  $s_i$  by convolution with  $h$ :  $s_i * h$
- determine the effect of  $L$  on  $s_i$  by multiplication with  $H$ :  $S_i \cdot H$

$$s_i * h \leftrightarrow S_i \cdot H$$

Since convolution is expensive for wide  $h$ , the multiplication may be cheaper

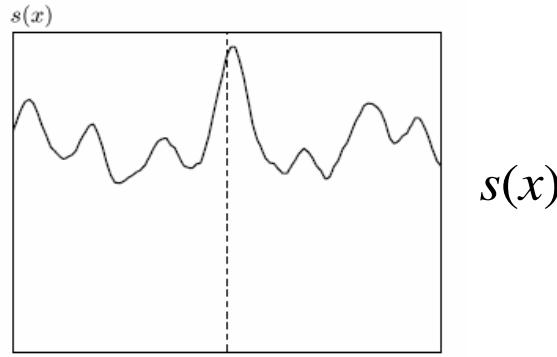
- but we need to perform the Fourier transforms of  $s_i$  and  $h$

# Fourier

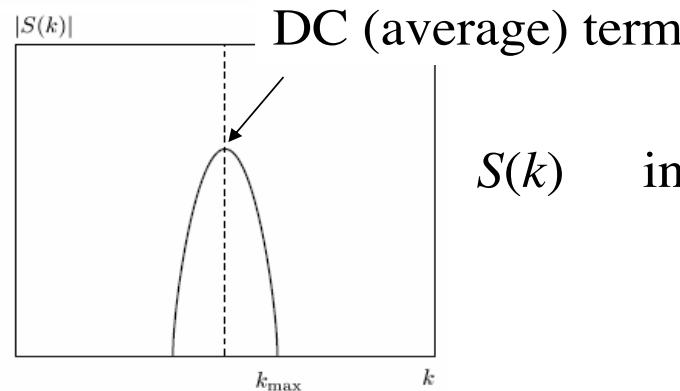
---

Theory developed by Joseph Fourier (1768-1830)

The Fourier transform of a signal  $s(x)$  yields its  
*frequency spectrum*  $S(k)$



$s(x)$  forward transform



$S(k)$  inverse transform

$$S(k) = F\{s(x)\} = \int_{-\infty}^{+\infty} s(x)e^{-2\pi ikx} dx$$

$$s(x) = F^{-1}\{S(k)\} = \int_{-\infty}^{+\infty} S(k)e^{2\pi ikx} dk$$

# Tutorial visual de FT

---

- <https://youtu.be/spUNpyF58BY>

# Generalização

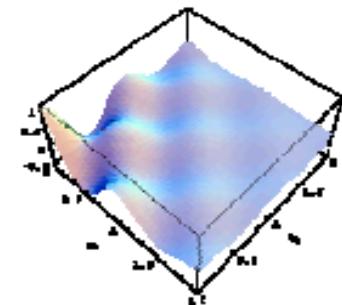
---

The Fourier transform generalizes to higher dimensions

Consider the 2D case:

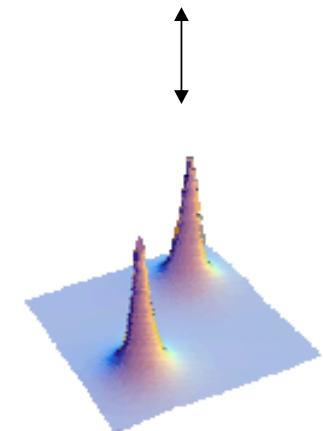
forward transform

$$S(k, l) = F\{s(x, y)\} = \int_{-\infty}^{+\infty} \int_{-\infty}^{+\infty} s(x, y) e^{-2\pi i(kx+ly)} dx dy$$



inverse transform

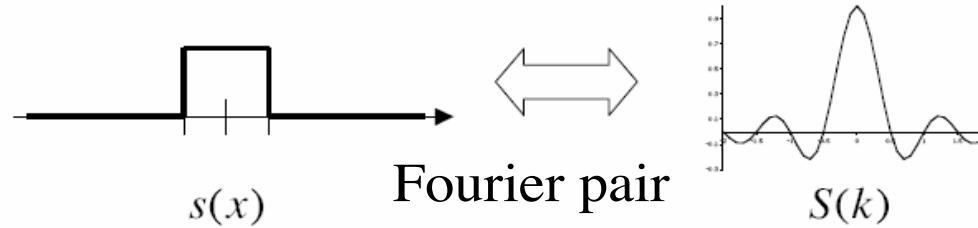
$$s(x, y) = F^{-1}\{S(k, l)\} = \int_{-\infty}^{+\infty} \int_{-\infty}^{+\infty} S(k, l) e^{2\pi i(kx+ly)} dk dl$$



# Exemplo

---

$$\begin{aligned}
 S(k) &= F\left\{A\Pi\left(\frac{x}{2L}\right)\right\} = \int_{-\infty}^{+\infty} A\Pi\left(\frac{x}{2L}\right)e^{-i2\pi kx}dx = \int_{-L}^{+L} Ae^{-i2\pi kx}dx \\
 &= -\frac{A}{2\pi ki}(e^{-i2\pi kL} - e^{i2\pi kL}) = \frac{A}{2\pi k}2\sin(2\pi kL) \\
 &= 2AL \operatorname{sinc}(2\pi kL)
 \end{aligned}$$



We see that a finite signal in the  $x$ -domain creates an infinite signal in the  $k$ -domain (the frequency domain)

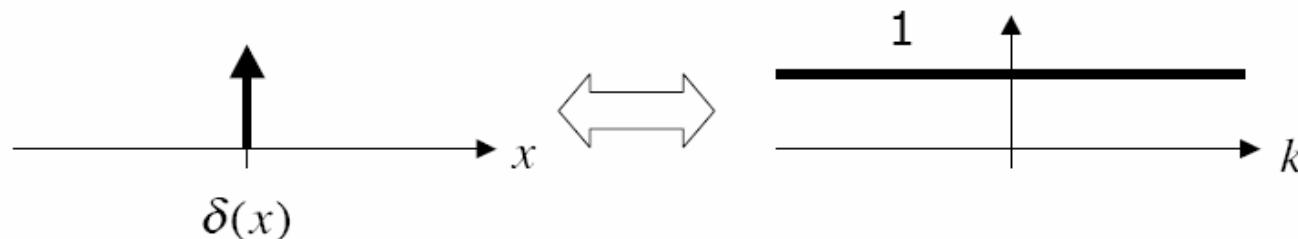
- the same is true vice versa

# Função impulso (Dirac)

---

For  $s(x)=\delta(x)$ :

$$S(k) = F\{\delta(x)\} = \int_{-\infty}^{+\infty} \delta(x) e^{-i2\pi kx} dx = e^{-i2\pi k0} = 1$$



Recall that the Dirac is an extremely thin rect function

- the frequency spectrum is therefore extremely broad (1 everywhere)

This illustrates a key feature of the Fourier Transform:

- the narrower the  $s(x)$ , the wider the  $S(k)$
- sharp objects need higher frequencies to represent that sharpness

# Propriedades

Scaling:



$$S(k) = 2AL \operatorname{sinc}(2\pi kL)$$

$$F\{s(ax)\} = \frac{1}{|a|} S\left(\frac{k}{a}\right)$$

$a > 1$  shrinks  $s$   
 $a < 1$  stretches  $s$

Linearity:  $F\{as_1(x) + bs_2(x)\} = F\{as_1(x)\} + F\{bs_2(x)\}$

Translation:  $F\{s(x - x_0)\} = S(k)e^{-2\pi i x_0 k}$

Convolution:  $F\{s_1(x) * s_2(x)\} = S_1(k) \cdot S_2(k)$  ← phase shift

$$F\{s_1(x) \cdot s_2(x)\} = S_1(k) * S_2(k)$$

# Influência na função de transferência

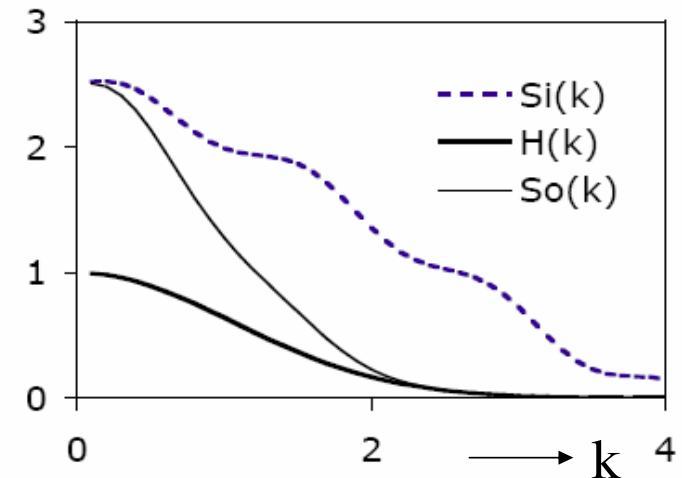
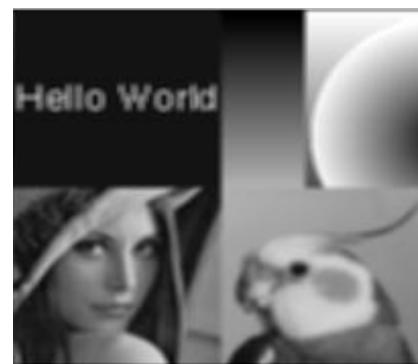
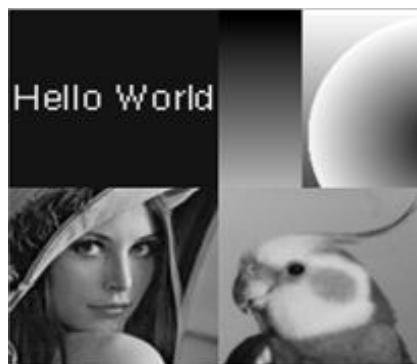
---

$$s_o(x) = \int_{-\infty}^{+\infty} S_i(k) e^{2\pi i k x} H(k) dk$$

$$s_o(x) = s_i(x) * h(x) \Leftrightarrow S_i(k) \cdot H(k) = S_o(k)$$

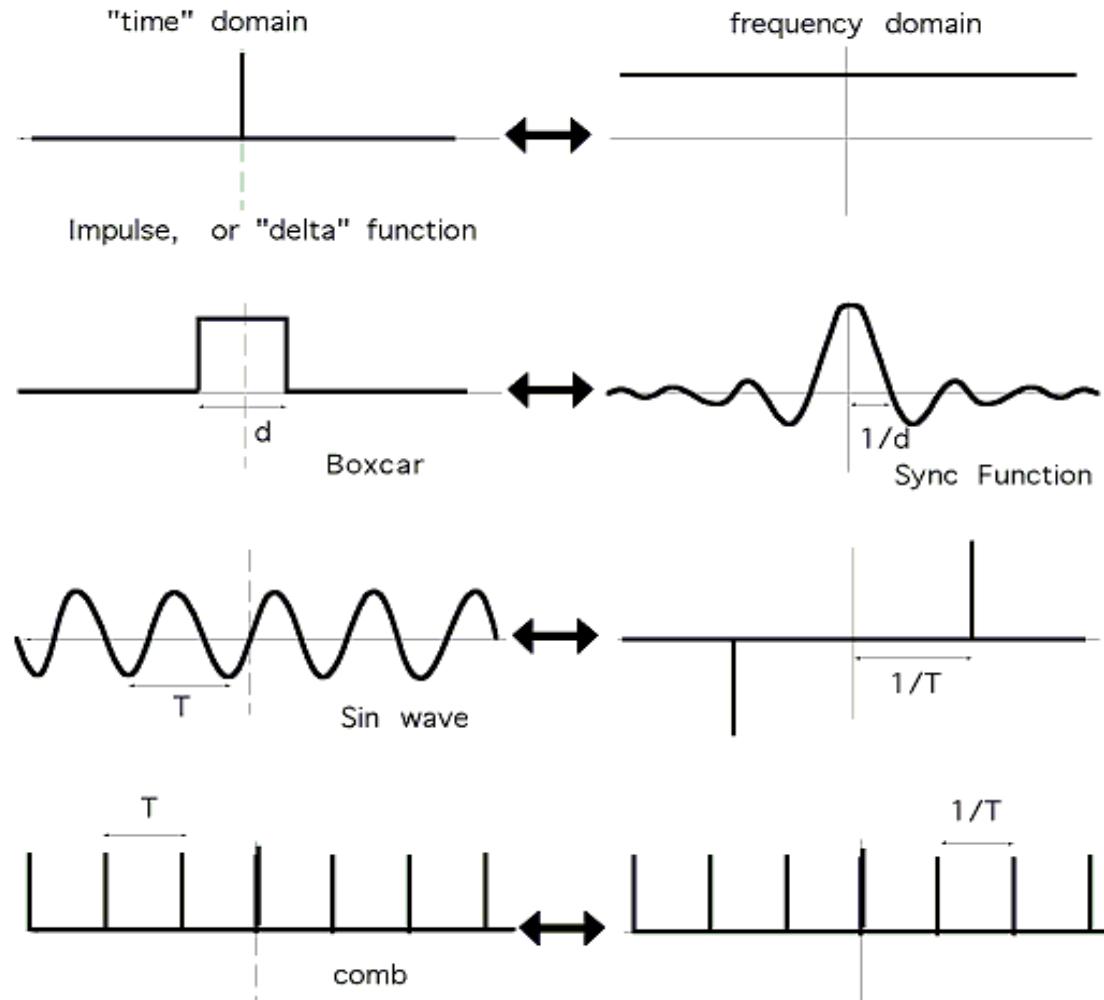
Let's look at a concrete example:

- $H$  is a *lowpass (blurring) filter*. it reduces the higher frequencies of  $S$  more than the lower ones



after application of  $H$

# Pares importantes



# Transformada discreta

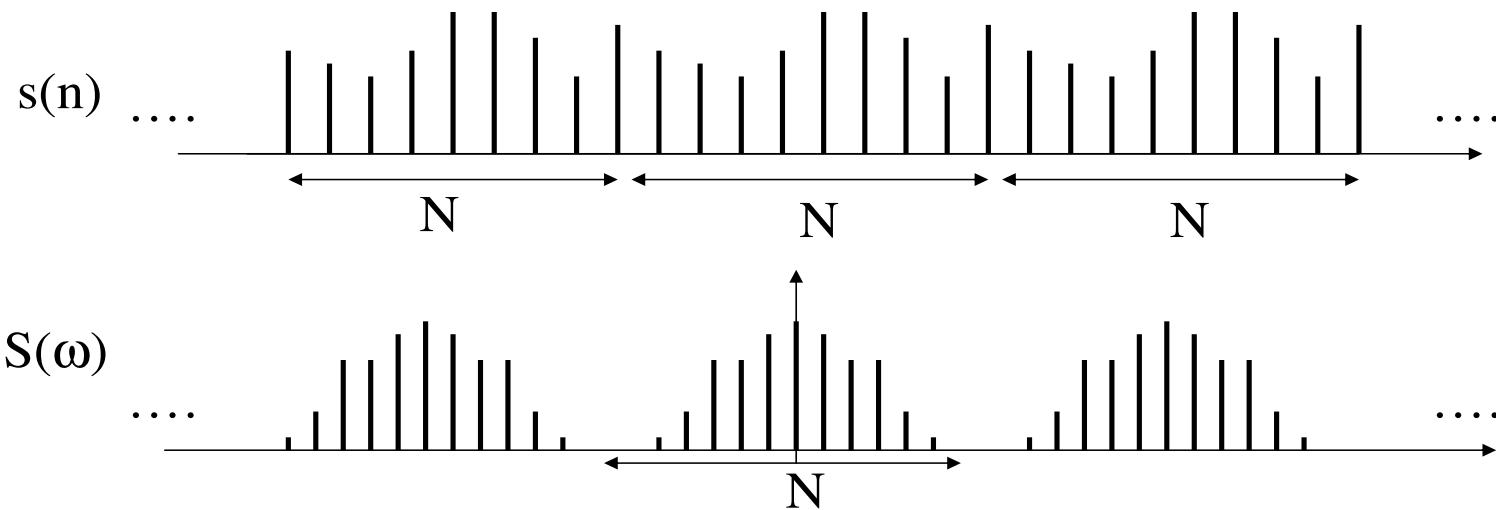
---

## Discrete Fourier Transform (DFT)

- assumes that the signal is discrete and finite

$$S(k) = \sum_{n=0}^{N-1} s(n) e^{\frac{-i2\pi kn}{N}} \quad s(n) = \frac{1}{N} \sum_{k=0}^{N-1} S(k) e^{\frac{i2\pi kn}{N}}$$

- now we have only  $N$  samples, and we can calculate  $N$  frequencies
- the frequency spectrum is now discrete, and it is periodic in  $N$



# Exemplo

---

The 2D transform:

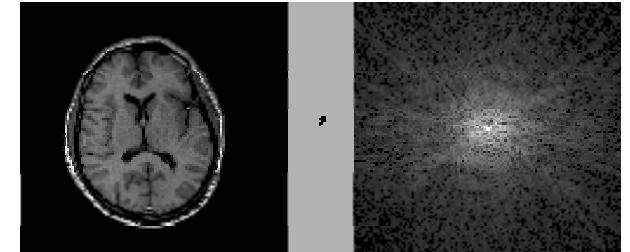
$$S(k,l) = \sum_{m=0}^{M-1} \sum_{n=0}^{N-1} s(n,m) e^{\frac{-i2\pi(kn+lm)}{NM}}$$

$$s(n,m) = \frac{1}{NM} \sum_{m=0}^{M-1} \sum_{n=0}^{N-1} S(k,l) e^{\frac{i2\pi(kn+lm)}{NM}}$$

Separability:

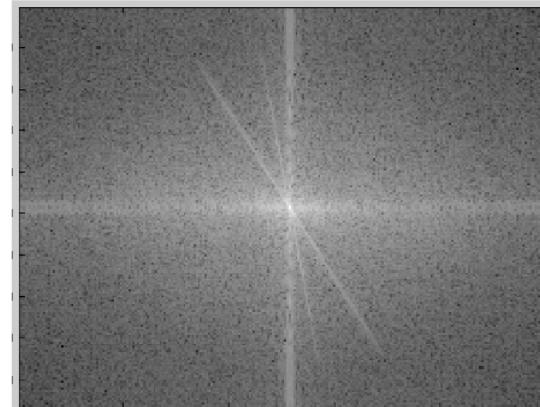
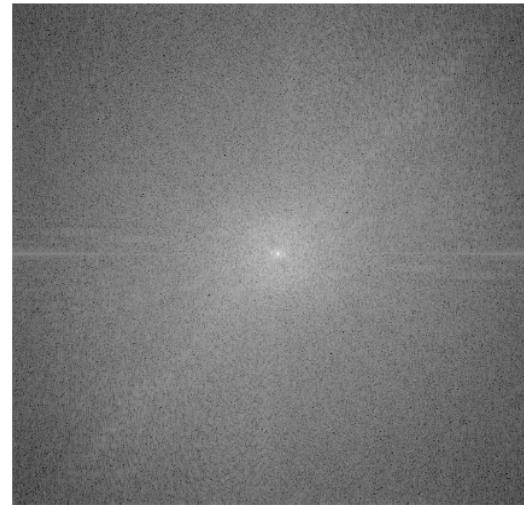
$$S(k,l) = \frac{1}{NM} \sum_{m=0}^{M-1} e^{\frac{-i2\pi lm}{M}} P(k,m) \quad \text{where } P(k,m) = \sum_{n=0}^{N-1} s(n,m) e^{\frac{-i2\pi kn}{N}}$$

$$s(n,m) = \frac{1}{NM} \sum_{l=0}^{M-1} e^{\frac{-i2\pi lm}{M}} p(n,l) \quad \text{where } p(n,l) = \sum_{k=0}^{N-1} S(n,m) e^{\frac{-i2\pi kn}{N}}$$



# Exemplos

---

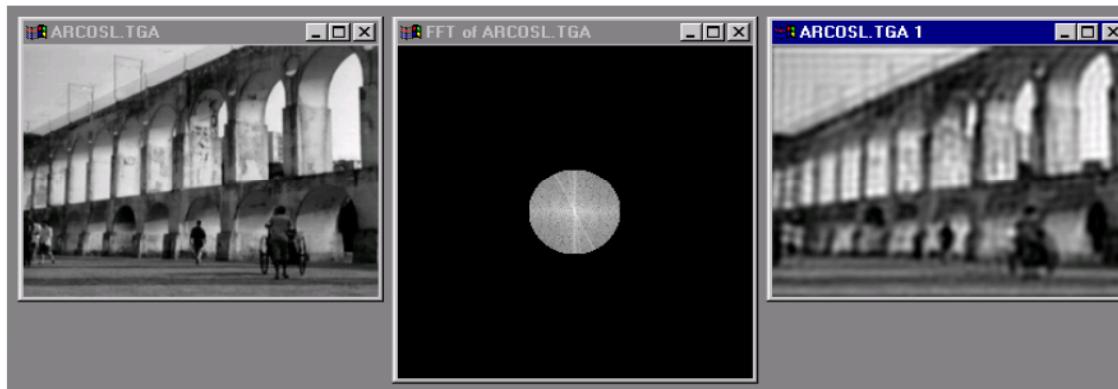


A. Efros, CMU

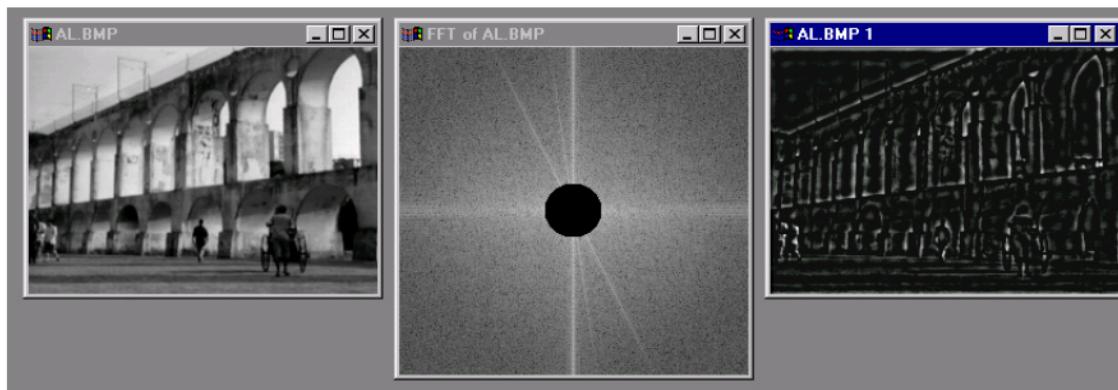
# Modificações no espectro

- (a) Lower frequencies (close to origin) give overall structure
- (b) Higher frequencies (periphery) give detail (sharp edges)

(a)

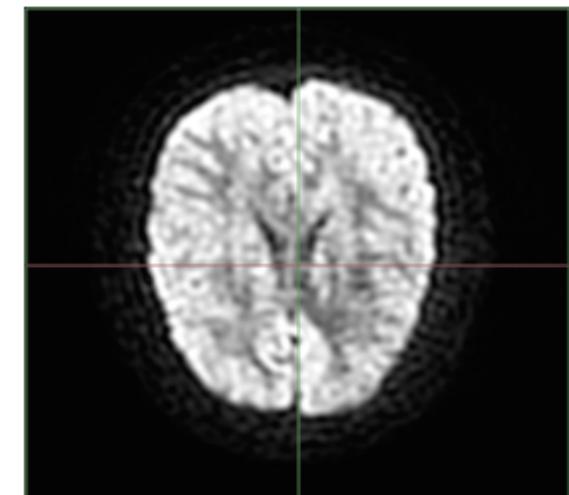
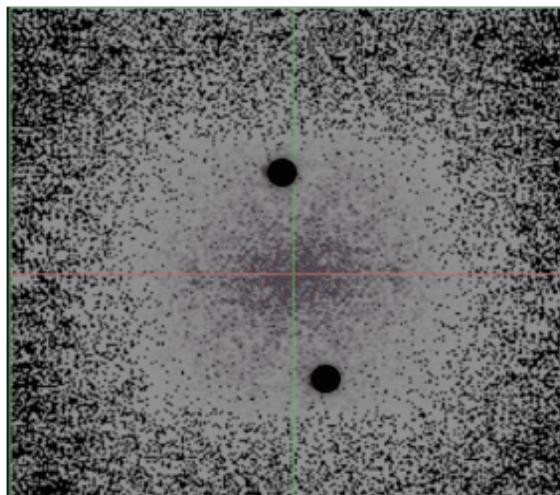
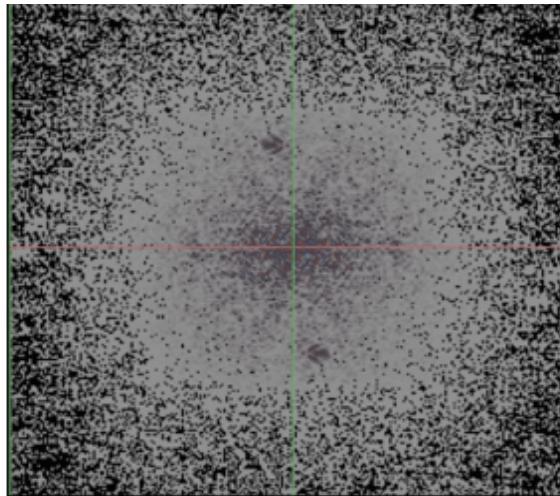
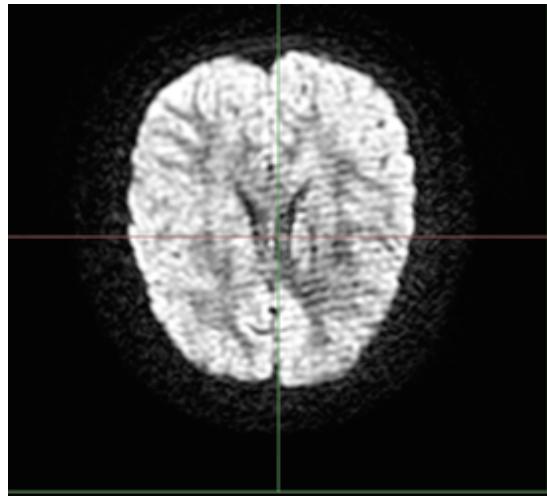


(b)



A. Efros, CMU

# Remoção de ruído



# Tarefa de casa

---

- Leitura do livro-texto até a seção 2.4