# Computer Vision

Spring 2006 15-385,-685

Instructor: S. Narasimhan

Wean 5403 T-R 3:00pm – 4:20pm

#### **Announcements**

- Homework 1 is due today in class.
- Homework 2 will be out later this evening (due in 2 weeks).
- Start homeworks early.
- Post questions on bboard.

## Image Processing and Filtering

#### Lecture #5

### Image as a Function

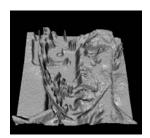
- We can think of an **image** as a function, f,
- f:  $R^2 \rightarrow R$ 
  - -f(x, y) gives the **intensity** at position (x, y)
  - Realistically, we expect the image only to be defined over a rectangle, with a finite range:
    - $f: [a,b] \times [c,d] \rightarrow [0,1]$
- A color image is just three functions pasted together. We can write this as a "vectorvalued" function:

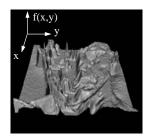
f(x,y) = 
$$\begin{vmatrix} r(x,y) \\ g(x,y) \\ b(x,y) \end{vmatrix}$$

## Image as a Function









### **Image Processing**

- Define a new image g in terms of an existing image f
  - We can transform either the domain or the range of f
- Range transformation:

$$g(x,y) = t(f(x,y))$$

What kinds of operations can this perform?

### **Image Processing**

• Some operations preserve the range but change the domain of *f*:

$$g(x,y) = f(t_x(x,y), t_y(x,y))$$

What kinds of operations can this perform?

• Still other operations operate on both the domain and the range of f .

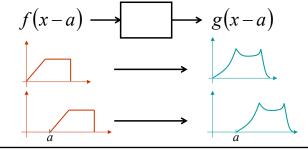
#### Linear Shift Invariant Systems (LSIS)

Linearity:

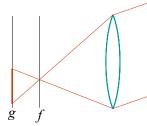
$$f_1 \longrightarrow g_1 \qquad f_2 \longrightarrow g_2$$

$$\alpha f_1 + \beta f_2 \longrightarrow \alpha g_1 + \beta g_2$$

Shift invariance:



## Example of LSIS



Defocused image ( g ) is a processed version of the focused image ( f )

Ideal lens is a LSIS 
$$f(x) \longrightarrow LSIS \longrightarrow g(x)$$

Linearity: Brightness variation

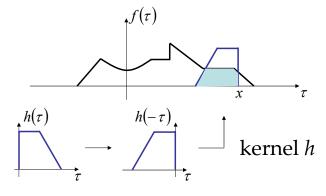
Shift invariance: Scene movement

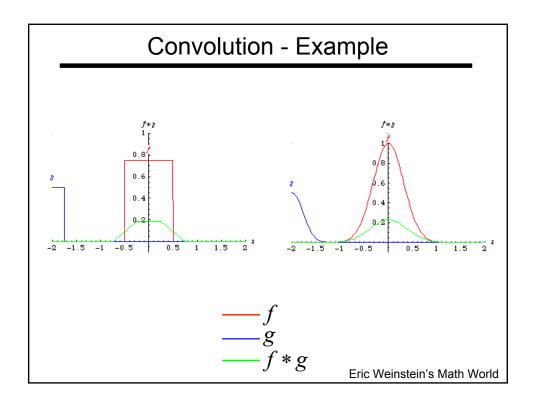
(not valid for lenses with non-linear distortions)

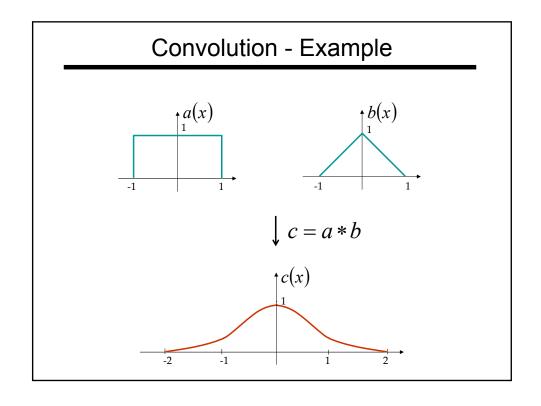
#### Convolution

LSIS is doing convolution; convolution is linear and shift invariant

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







#### Convolution Kernel - Impulse Response

$$f \longrightarrow h \longrightarrow g$$

$$\Rightarrow g$$
  $g = f * h$ 

• What h will give us g = f?

Dirac Delta Function (Unit Impulse)

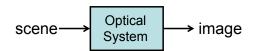
 $\begin{array}{c|c}
 & \delta(x) \\
 & & \\
 & & \\
 & & \\
 & \varepsilon \to 0
\end{array}$ 

Sifting property:

$$\int_{-\infty}^{\infty} f(x)\delta(x)dx = \int_{-\infty}^{\infty} f(0)\delta(x)dx$$
$$= f(0)\int_{-\infty}^{\infty} \delta(x)dx = f(0)$$

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

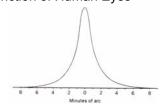
### Point Spread Function



- Ideally, the optical system should be a Dirac delta function.
- · However, optical systems are never ideal.

$$\begin{array}{ccc}
\delta(x) & \longrightarrow & \text{Optical} \\
\text{point source} & & \text{System} & & \text{point spread function}
\end{array}$$

· Point spread function of Human Eyes



## Point Spread Function







normal vision

myopia

hyperopia



astigmatism

Images by Richmond Eye Associates

## **Properties of Convolution**

Commutative

$$a * b = b * a$$

Associative

$$(a*b)*c = a*(b*c)$$

Cascade system

$$f \longrightarrow h_1 \longrightarrow h_2 \longrightarrow g$$

$$= f \longrightarrow h_1 * h_2 \longrightarrow g$$

$$= f \longrightarrow h_2 * h_1 \longrightarrow g$$

#### How to Represent Signals?

Option 1: Taylor series represents any function using polynomials.

$$f(x) = f(\alpha) + f'(\alpha)(x - \alpha) + \frac{f''(\alpha)}{2!}$$
$$(x - \alpha)^{2} + \frac{f^{(3)}(\alpha)}{3!}(x - \alpha)^{3} + \dots + \frac{f^{(n)}(\alpha)}{n!}(x - \alpha)^{n} + \dots$$

- Polynomials are not the best unstable and not very physically meaningful.
- Easier to talk about "signals" in terms of its "frequencies" (how fast/often signals change, etc).

#### Jean Baptiste Joseph Fourier (1768-1830)

- Had crazy idea (1807):
- Any periodic function can be rewritten as a weighted sum of Sines and Cosines of different frequencies.
- Don't believe it?
  - Neither did Lagrange, Laplace, Poisson and other big wigs
  - Not translated into English until 1878!
- But it's true!
  - called Fourier Series
  - Possibly the greatest tool used in Engineering

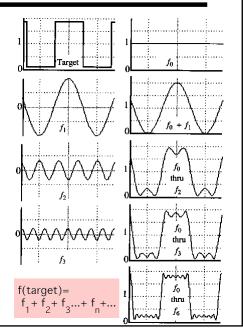


#### A Sum of Sinusoids

· Our building block:

$$A \sin(\omega x + \phi)$$

- Add enough of them to get any signal f(x) you want!
- How many degrees of freedom?
- · What does each control?
- Which one encodes the coarse vs. fine structure of the signal?



#### **Fourier Transform**

• We want to understand the frequency  $\omega$  of our signal. So, let's reparametrize the signal by  $\omega$  instead of x:

$$f(x)$$
 Fourier Transform  $\rightarrow$   $F(\omega)$ 

- For every  $\omega$  from 0 to inf,  $F(\omega)$  holds the amplitude A and phase  $\phi$  of the corresponding sine  $A \sin(\omega x + \phi)$ 
  - How can F hold both? Complex number trick!

$$F(\omega) = R(\omega) + iI(\omega)$$

$$A = \pm \sqrt{R(\omega)^2 + I(\omega)^2} \qquad \phi = \tan^{-1} \frac{I(\omega)}{R(\omega)}$$

 $F(\omega)$  Inverse Fourier Transform f(x)

## Time and Frequency

• example :  $g(t) = \sin(2pf t) + (1/3)\sin(2p(3f) t)$ 

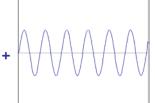


## Time and Frequency

• example :  $g(t) = \sin(2pf t) + (1/3)\sin(2p(3f) t)$ 

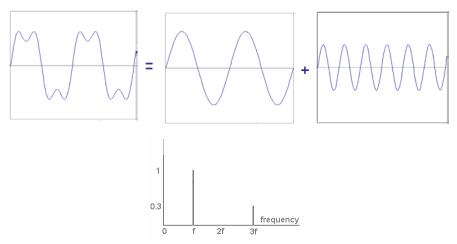






## Frequency Spectra

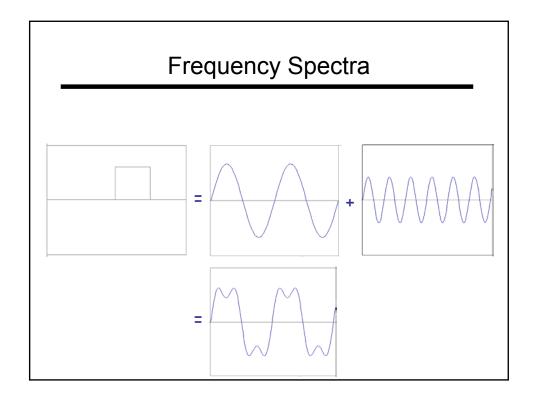
• example :  $g(t) = \sin(2pf t) + (1/3)\sin(2p(3f) t)$ 

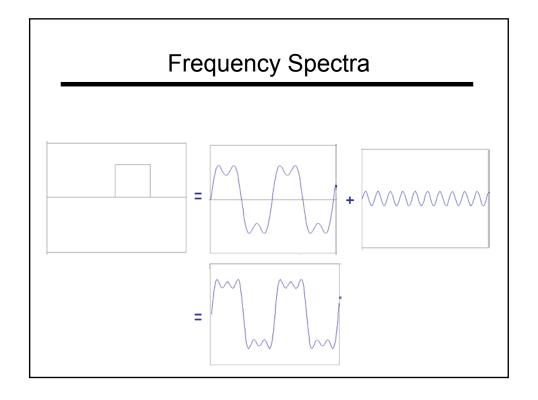


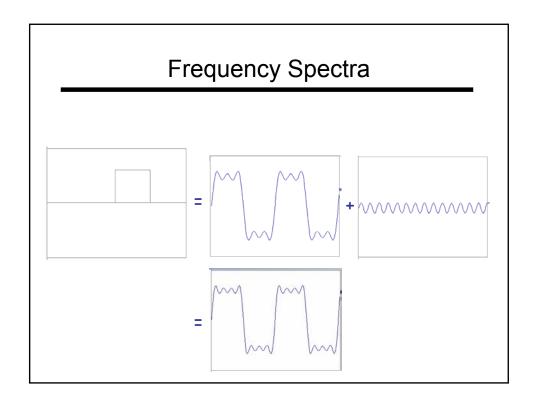
## Frequency Spectra

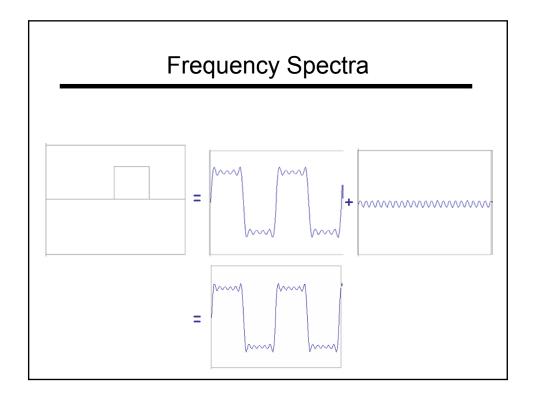
Usually, frequency is more interesting than the phase

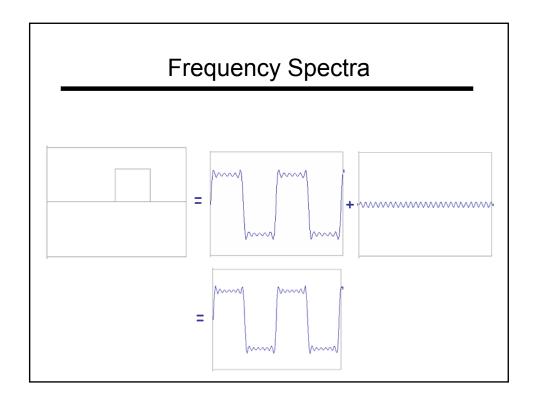


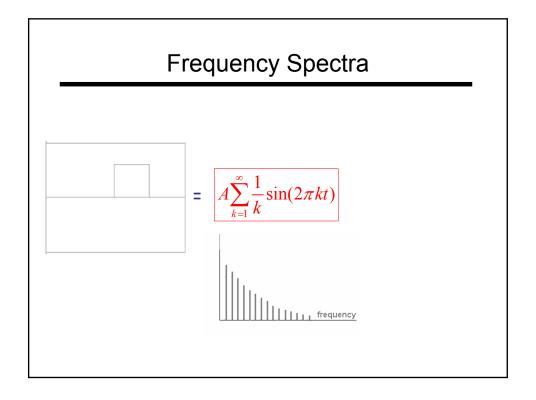


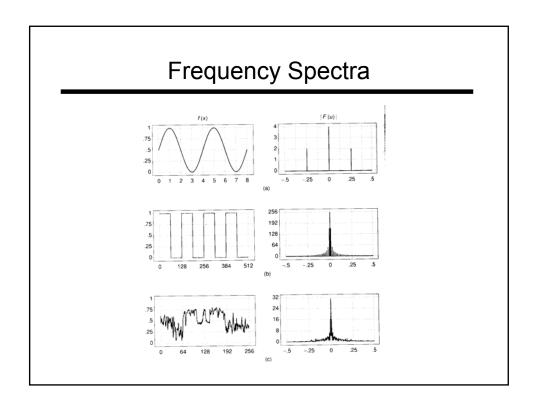


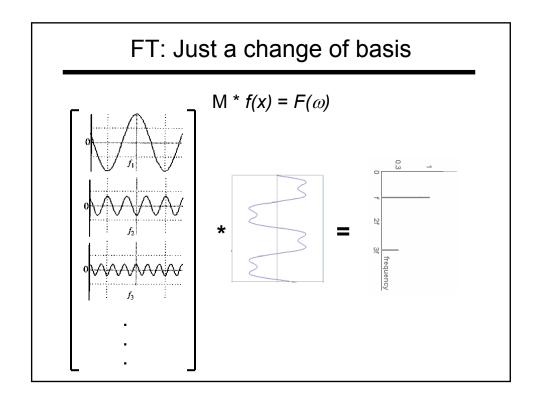






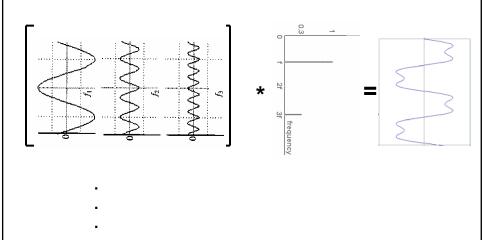






## IFT: Just a change of basis

$$\mathsf{M}^{-1} * F(\omega) = f(x)$$



## Fourier Transform – more formally

Represent the signal as an infinite weighted sum of an infinite number of sinusoids

$$F(u) = \int_{-\infty}^{\infty} f(x)e^{-i2\pi ux} dx$$

Note: 
$$e^{ik} = \cos k + i \sin k$$
  $i = \sqrt{-1}$ 

Arbitrary function  $\longrightarrow$  Single Analytic Expression Spatial Domain (x)  $\longrightarrow$  Frequency Domain (u) (Frequency Spectrum F(u))

Inverse Fourier Transform (IFT)

$$f(x) = \int_{-\infty}^{\infty} F(u)e^{i2\pi ux} dx$$

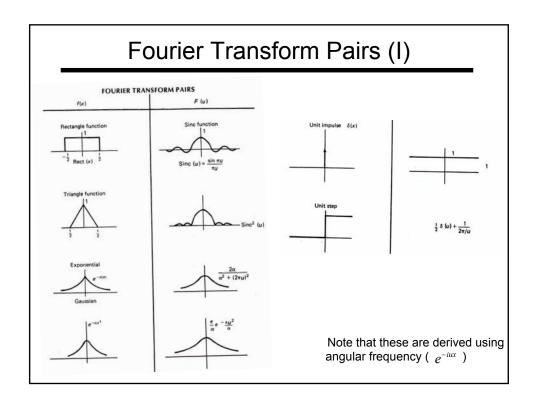
#### **Fourier Transform**

· Also, defined as:

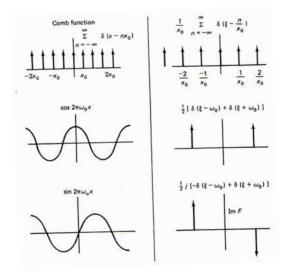
$$F(u) = \int_{-\infty}^{\infty} f(x)e^{-iux} dx$$
Note:  $e^{ik} = \cos k + i\sin k$   $i = \sqrt{-1}$ 

• Inverse Fourier Transform (IFT)

$$f(x) = \frac{1}{2\pi} \int_{-\infty}^{\infty} F(u) e^{iux} dx$$



## Fourier Transform Pairs (I)



Note that these are derived using angular frequency ( $e^{-iux}$ )

#### Fourier Transform and Convolution

Let 
$$g = f * h$$
  
Then  $G(u) = \int_{-\infty}^{\infty} g(x)e^{-i2\pi ux}dx$   
 $= \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} f(\tau)h(x-\tau)e^{-i2\pi u\tau}d\tau d\tau$   
 $= \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} \left[f(\tau)e^{-i2\pi u\tau}d\tau\right]h(x-\tau)e^{-i2\pi u(x-\tau)}dx$   
 $= \int_{-\infty}^{\infty} \left[f(\tau)e^{-i2\pi u\tau}d\tau\right]\int_{-\infty}^{\infty} \left[h(x')e^{-i2\pi ux'}dx'\right]$   
 $= F(u)H(u)$ 

Convolution in spatial domain

#### Fourier Transform and Convolution

Spatial Domain (x) Frequency Domain (u)
$$g = f * h \longleftrightarrow G = FH$$

$$g = fh \longleftrightarrow G = F * H$$

So, we can find g(x) by Fourier transform

$$g = f * h$$

$$\downarrow \text{FT}$$

$$\downarrow G = F \times H$$

### **Properties of Fourier Transform**

Spatial Domain (x)

Linearity 
$$c_1 f(x) + c_2 g(x)$$

Scaling 
$$f(ax)$$

Shifting 
$$f(x-x_0)$$

Symmetry 
$$F(x)$$

Conjugation 
$$f^*(x)$$

Convolution 
$$f(x)*g(x)$$

Differentiation 
$$\frac{d^n f(x)}{dx^n}$$

Frequency Domain (u)

$$c_1F(u)+c_2G(u)$$

$$\frac{1}{|a|}F\left(\frac{u}{a}\right)$$

$$e^{-i2\pi u x_0}F(u)$$

$$f(-u)$$

$$F^*(-u)$$

$$(i2\pi u)^n F(u)$$

Note that these are derived using frequency (  $e^{-i2\pi \imath x}$  )

## **Properties of Fourier Transform**

Parseval's theorem:
$$\int_{-\infty}^{\infty} |f(x)|^2 dx = \int_{-\infty}^{\infty} |F(\xi)|^2 d\xi$$

$$\int_{-\infty}^{\infty} f(x)g^*(x) dx = \int_{-\infty}^{\infty} F(\xi)G^*(\xi) d\xi$$

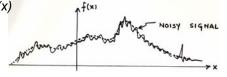
$$f(x) \qquad F(\xi)$$
Real(R)
Real part even (RE)
Imaginary part odd (IO)

Imaginary (I)
RE,IO
RE,IE
RE
RE
RO
IO
IE
RE
RO
IO
IE
RO
Complex even (CE)
CO
CO
CO

## Example use: Smoothing/Blurring

We want a smoothed function of f(x)

$$g(x) = f(x) * h(x)$$



 $h(x) \uparrow$ 

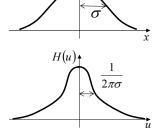
· Let us use a Gaussian kernel

$$h(x) = \frac{1}{\sqrt{2\pi}\sigma} \exp\left[-\frac{1}{2}\frac{x^2}{\sigma^2}\right]$$

Then

$$H(u) = \exp\left[-\frac{1}{2}(2\pi u)^2 \sigma^2\right]$$

$$G(u) = F(u)H(u)$$

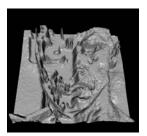


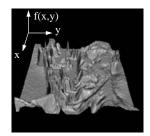
H(u) attenuates high frequencies in F(u) (Low-pass Filter)!

## Image as a Discrete Function









## **Digital Images**

#### The scene is

- projected on a 2D plane,
- sampled on a regular grid, and each sample is
- quantized (rounded to the nearest integer)

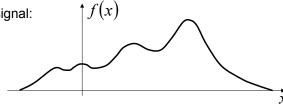
$$f(i, j) = \text{Quantize}\{f(i\Delta, j\Delta)\}$$

#### Image as a matrix

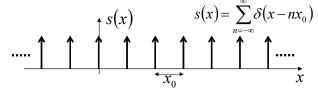
	$\xrightarrow{j}$							
i	62	79	23	119	120	105	4	0
	10	10	9	62	12	78	34	0
¥	10	58	197	46	46	0	0	48
	176	135	5	188	191	68	0	49
	2	1	1	29	26	37	0	77
	0	89	144	147	187	102	62	208
	255	252	0	166	123	62	0	31
	166	63	127	17	1	0	99	30

### Sampling Theorem

Continuous signal:



Shah function (Impulse train):

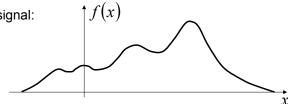


Sampled function:

$$f_s(x) = f(x)s(x) = f(x)\sum_{n=-\infty}^{\infty} \delta(x - nx_0)$$

### Sampling Theorem

Continuous signal:



Shah function (Impulse train):

$$S(x) = \sum_{n=-\infty} S(x - nx_0)$$

$$X(x) = \sum_{n=-\infty} S(x - nx_0)$$

Sampled function:

$$f_s(x) = f(x)s(x) = f(x)\sum_{n=-\infty}^{\infty} \delta(x - nx_0)$$

### Sampling Theorem

Sampled function:

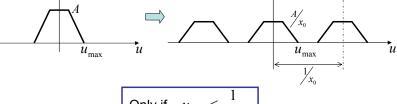
Sampling 
$$\frac{1}{x_0}$$

$$F_S(u) = F(u) * S(u) = F(u) * \frac{1}{x_0} \sum_{n=-\infty}^{\infty} \delta(x - nx_0)$$

$$F_S(u) = F(u) * S(u) = F(u) * \frac{1}{x_0} \sum_{n=-\infty}^{\infty} \delta\left(u - \frac{n}{x_0}\right)$$

$$F(u)$$

$$F(u)$$



Only if  $u_{\text{max}} \le \frac{1}{2x_0}$ 

## **Nyquist Theorem**

If 
$$u_{\text{max}} > \frac{1}{2x_0}$$

$$u_{\text{max}} > \frac{1}{2x_0}$$
Aliasing

When can we recover F(u) from  $F_s(u)$  ?

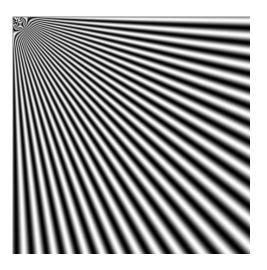
Only if 
$$u_{\text{max}} \le \frac{1}{2x_0}$$
 (Nyquist Frequency)

We can use  $C(u) = \begin{cases} x_0 & |u| < \frac{1}{2}x_0 \\ 0 & \text{otherwise} \end{cases}$ 

Then 
$$F(u) = F_S(u)C(u)$$
 and  $f(x) = IFT[F(u)]$ 

Sampling frequency must be greater than  $\,2u_{
m max}$ 

## Aliasing



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## Next Class

- Image Processing and Filtering (continued)
- Horn, Chapter 6