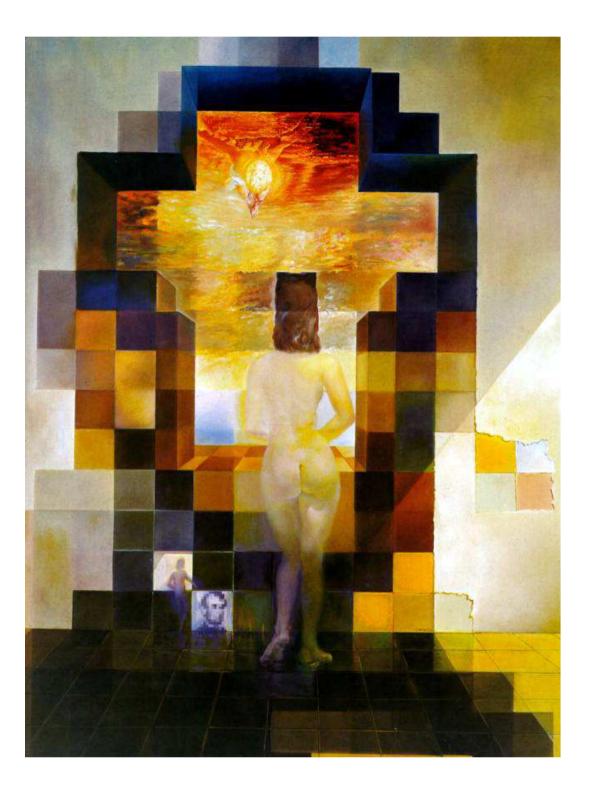
The Frequency Domain



Somewhere in Cinque Terre, May 2005

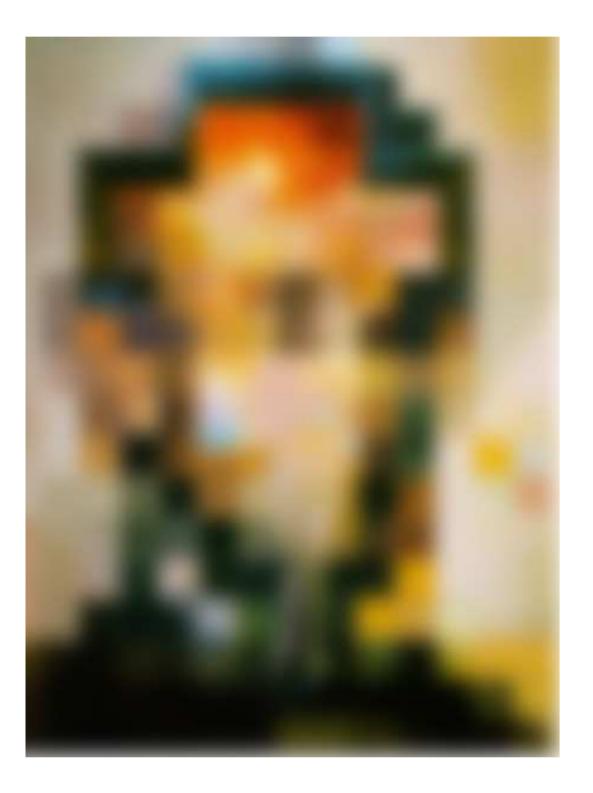
15-463: Computational Photography Alexei Efros, CMU, Fall 2011

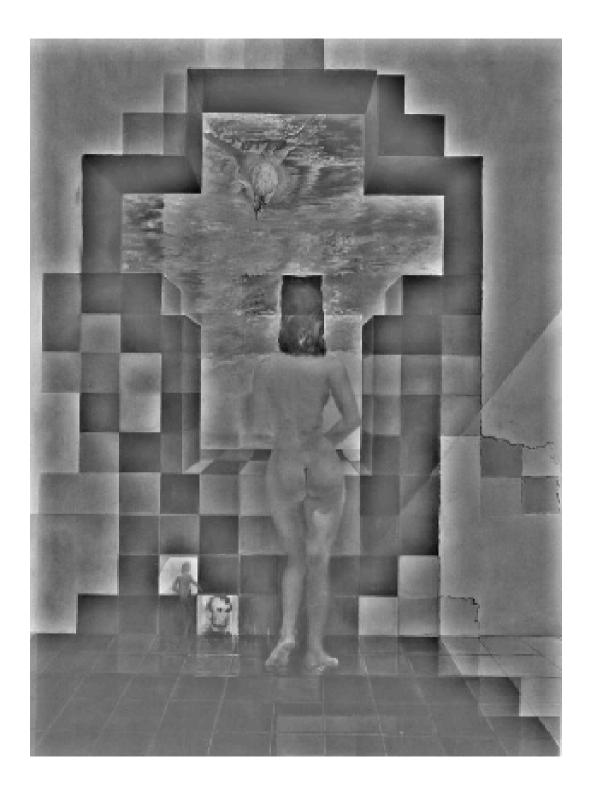
Many slides borrowed from Steve Seitz



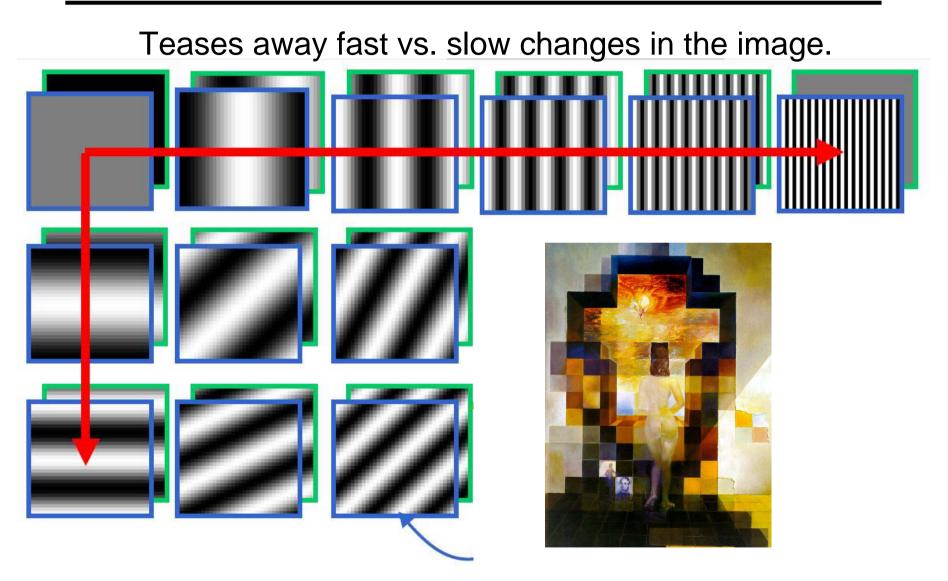
Salvador Dali

"Gala Contemplating the Mediterranean Sea, which at 30 meters becomes the portrait of Abraham Lincoln", 1976





A nice set of basis



This change of basis has a special name...

Jean Baptiste Joseph Fourier (1768-1830)

had crazy idea (1807

Any univariate function ca be rewritten as a weighted sum of sines and cosines different frequencies.

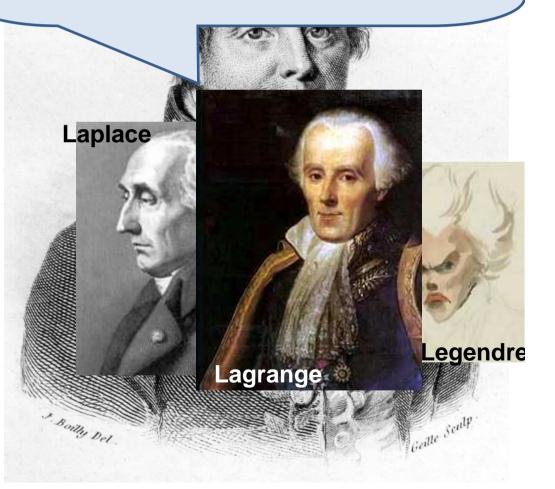
Don't believe it?

- Neither did Lagrange, Laplace, Poisson and other big wigs
- Not translated into English until 1878!

But it's (mostly) true!

- called Fourier Series
- there are some subtle restrictions

...the manner in which the author arrives at these equations is not exempt of difficulties and...his analysis to integrate them still leaves something to be desired on the score of generality and even rigour.



A sum of sines

Our building block:

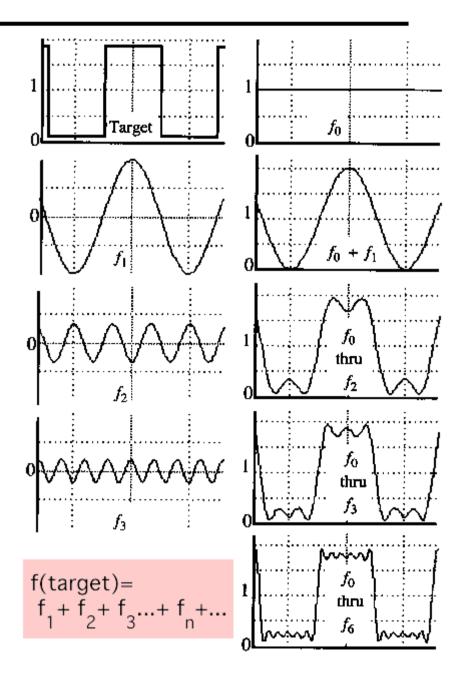
 $A\sin(\omega x + \phi)$

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 ω of our signal. So, let's reparametrize the signal by ω instead of *x*:



For every ω from 0 to inf, $F(\omega)$ holds the amplitude A and phase ϕ of the corresponding sine $A \sin(\omega x + \phi)$

• How can *F* hold both? Complex number trick!

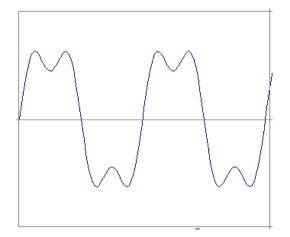
$$F(\omega) = R(\omega) + iI(\omega)$$
$$A = \pm \sqrt{R(\omega)^2 + I(\omega)^2} \qquad \phi = \tan^{-1} \frac{I(\omega)}{R(\omega)}$$

We can always go back:

$$\begin{array}{c} F(\omega) \longrightarrow & \text{Inverse Fourier} \\ & \text{Transform} \end{array} \longrightarrow f(x) \end{array}$$

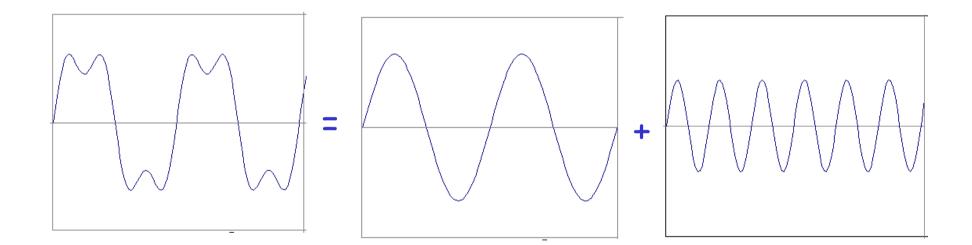
Time and Frequency

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

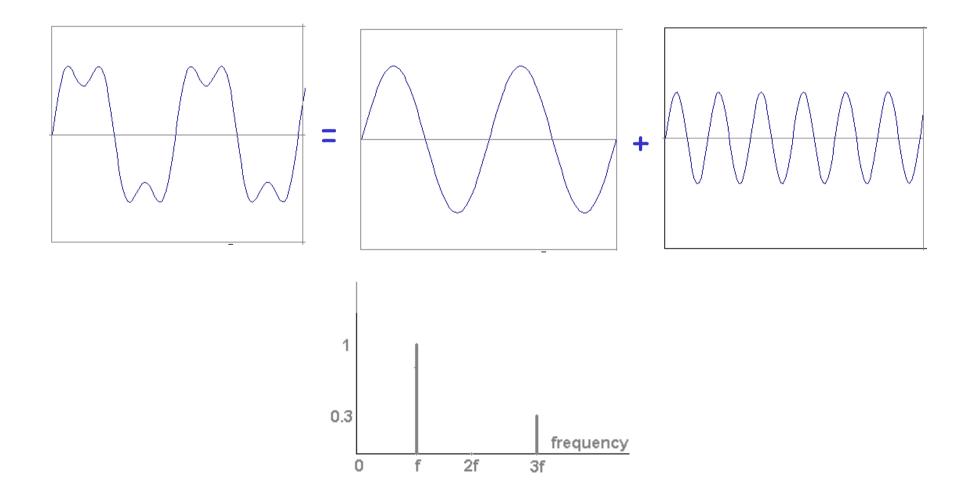


Time and Frequency

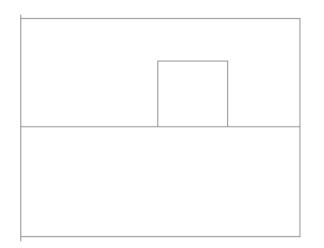
example : g(t) = sin(2pf t) + (1/3)sin(2p(3f) t)

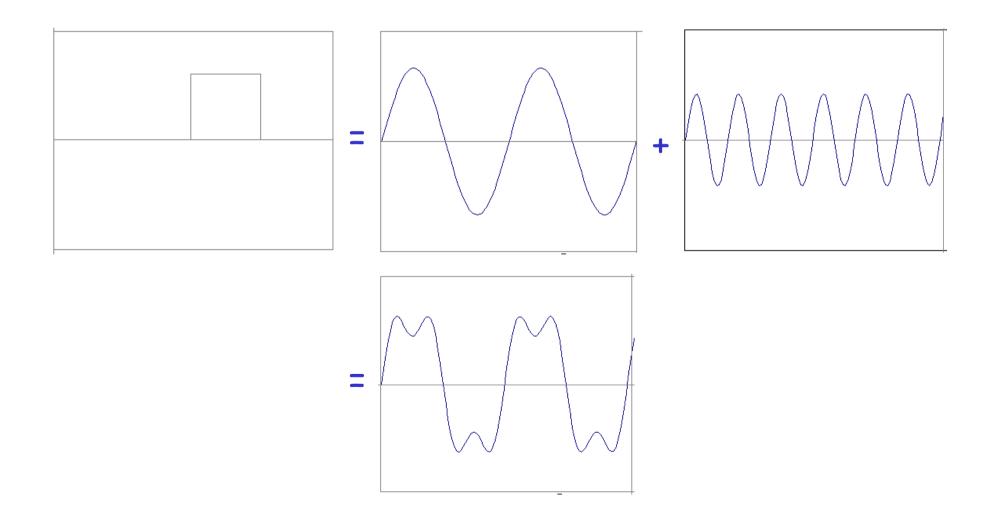


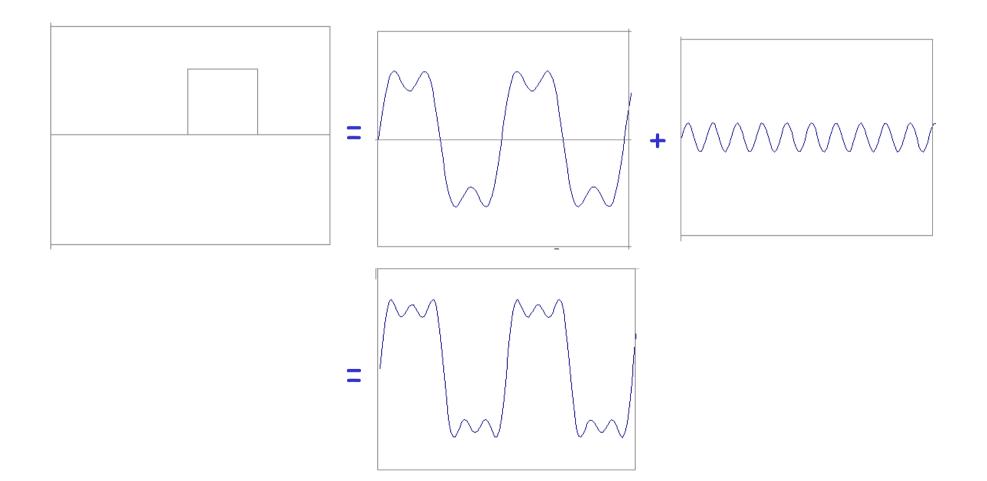
example : g(t) = sin(2pf t) + (1/3)sin(2p(3f) t)

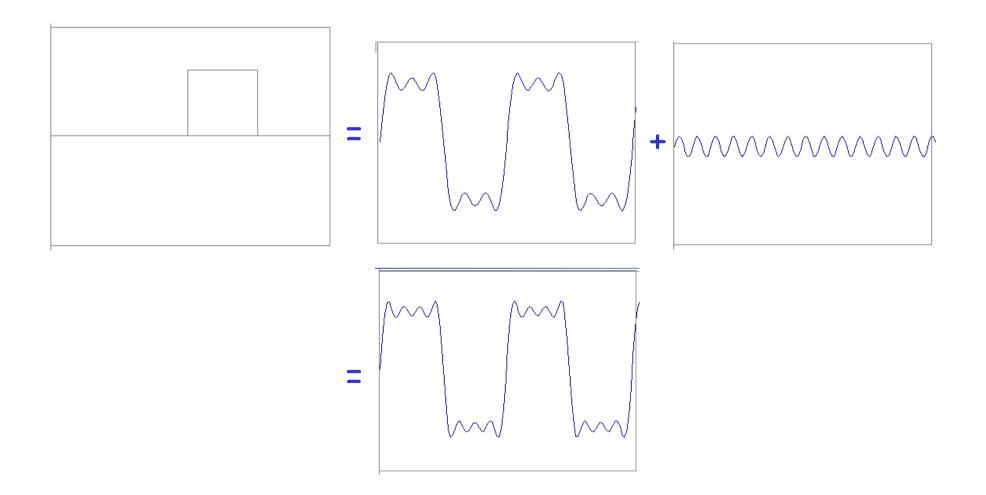


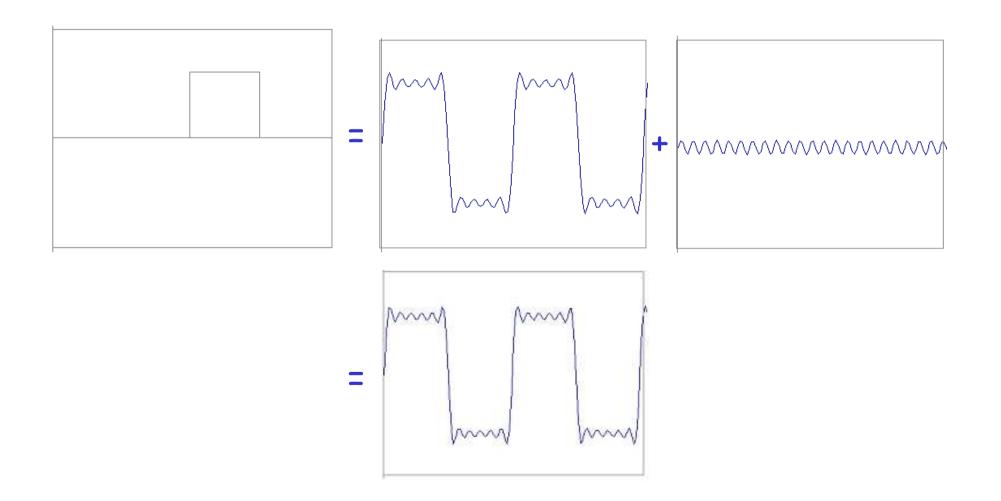
Usually, frequency is more interesting than the phase

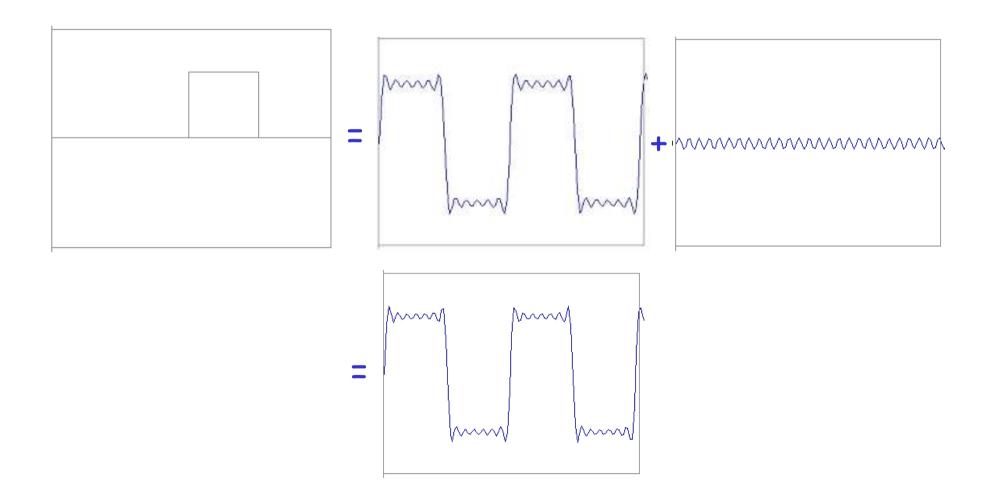


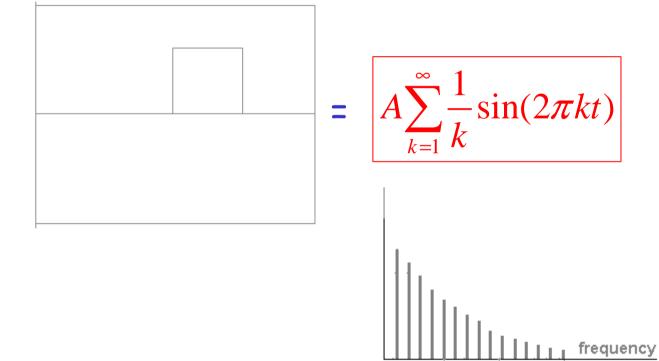


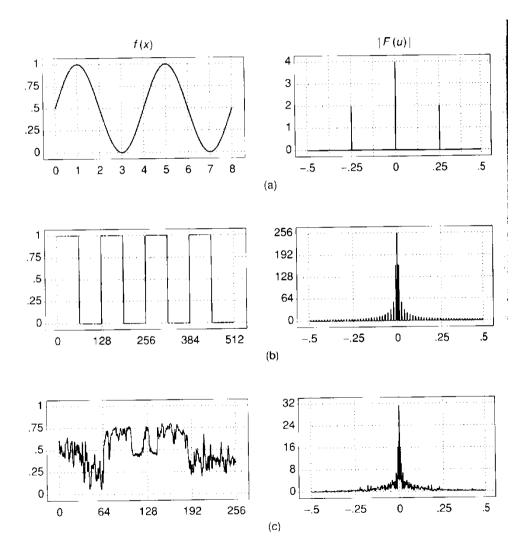




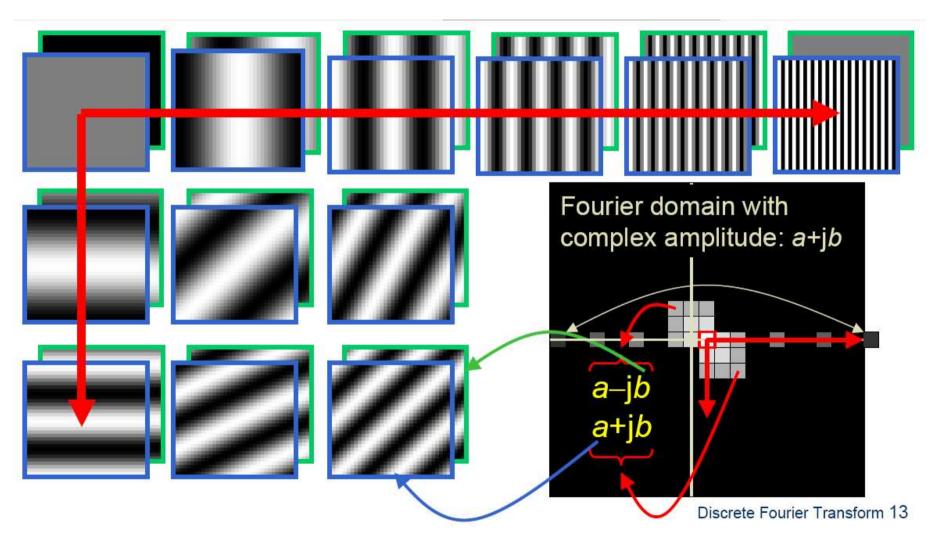






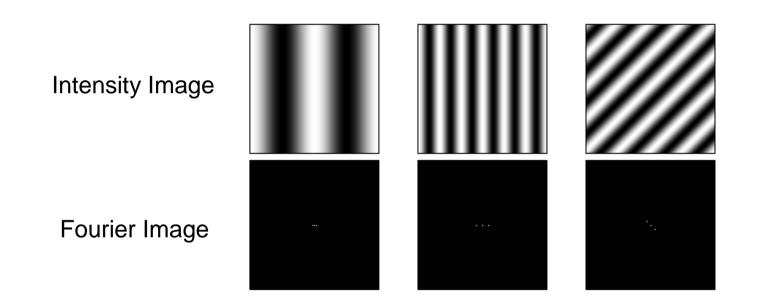


Extension to 2D



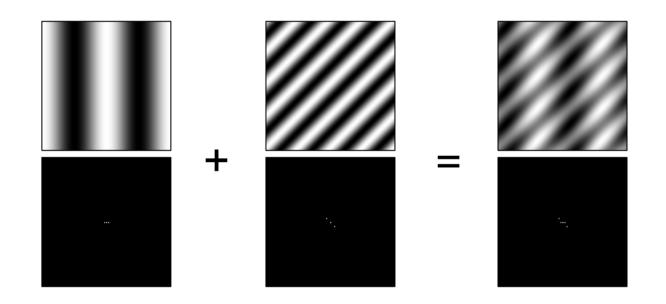
in Matlab, check out: imagesc(log(abs(fftshift(fft2(im)))));

Fourier analysis in images



http://sharp.bu.edu/~slehar/fourier/fourier.html#filtering

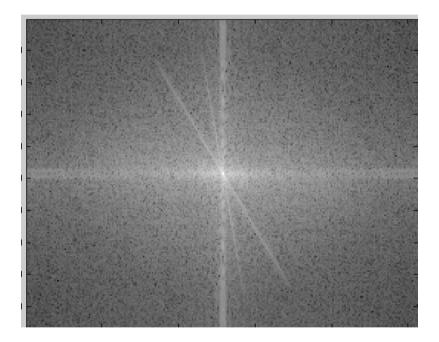
Signals can be composed



http://sharp.bu.edu/~slehar/fourier/fourier.html#filtering More: http://www.cs.unm.edu/~brayer/vision/fourier.html

Man-made Scene

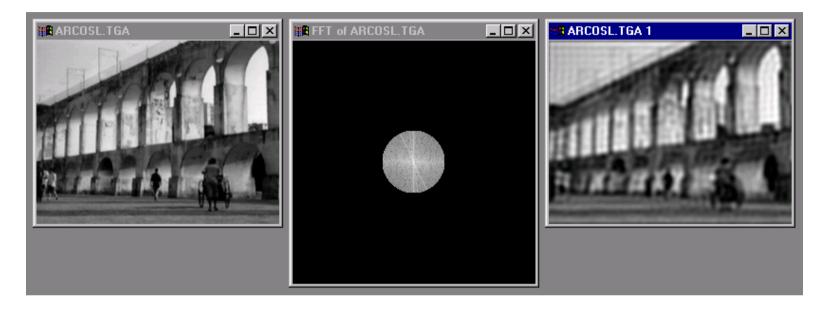


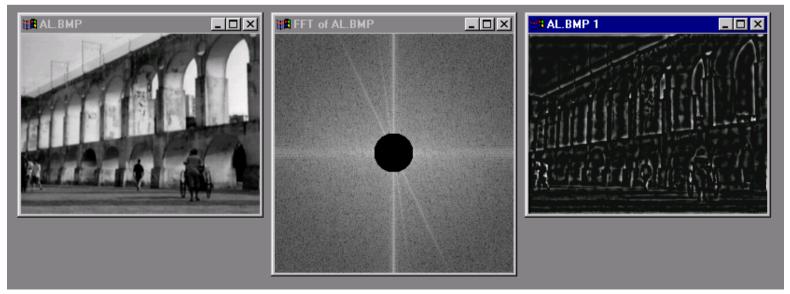


Can change spectrum, then reconstruct



Low and High Pass filtering





The Convolution Theorem

The greatest thing since sliced (banana) bread!

- The Fourier transform of the convolution of two functions is the product of their Fourier transforms F[g * h] = F[g]F[h]
- The inverse Fourier transform of the product of two Fourier transforms is the convolution of the two inverse Fourier transforms

$$F^{-1}[gh] = F^{-1}[g] * F^{-1}[h]$$

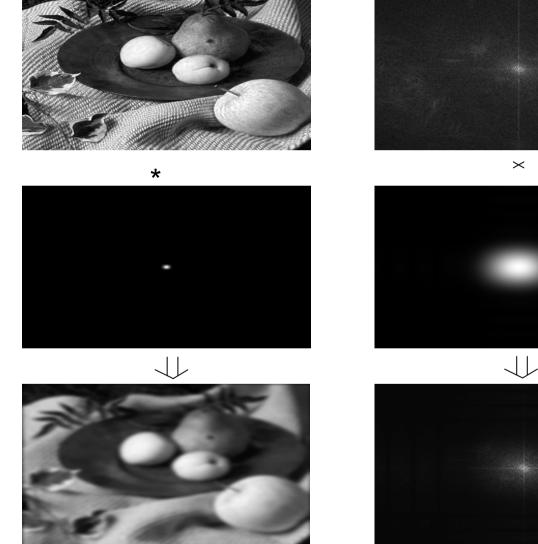
• **Convolution** in spatial domain is equivalent to **multiplication** in frequency domain!

2D convolution theorem example



h(x,y)

g(x,y)



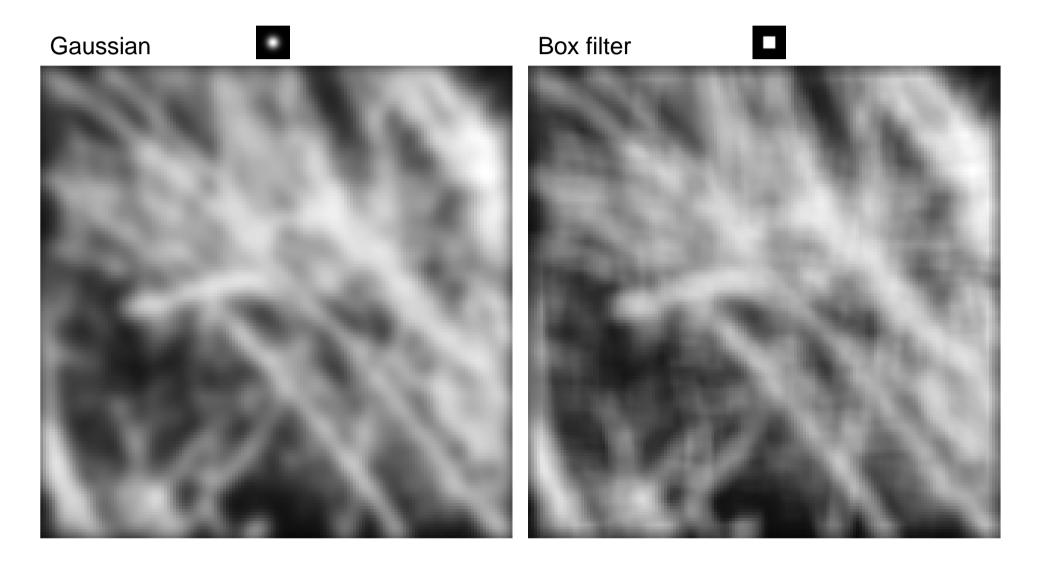
 $|F(s_x, s_y)|$

 $|H(s_x, s_y)|$

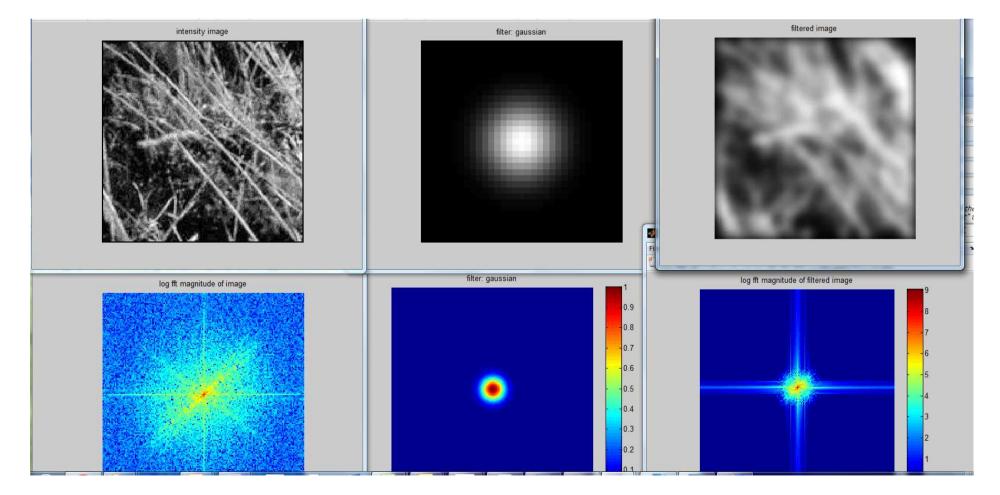
 $|G(s_x, s_y)|$

Filtering

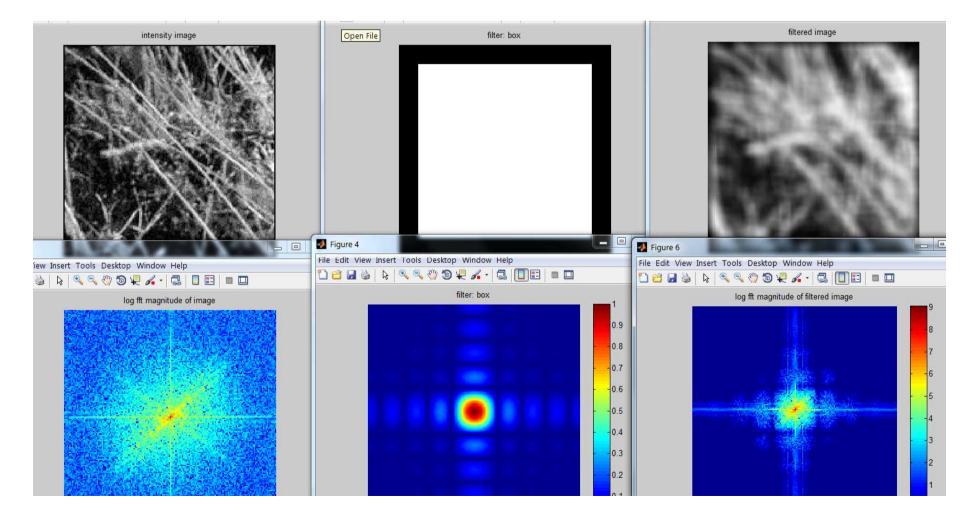
Why does the Gaussian give a nice smooth image, but the square filter give edgy artifacts?



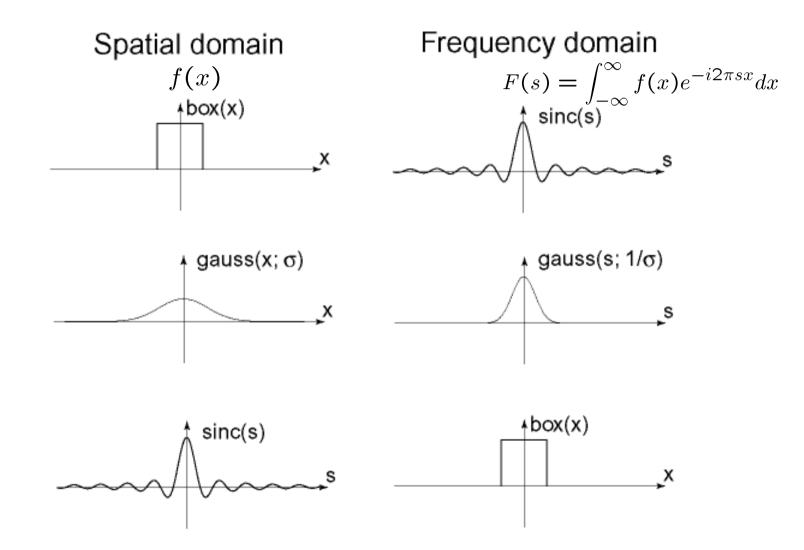
Gaussian



Box Filter

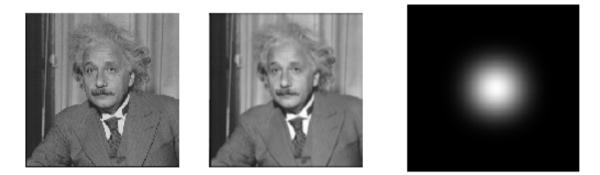


Fourier Transform pairs

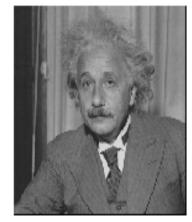


Low-pass, Band-pass, High-pass filters

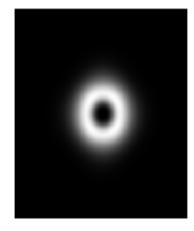
low-pass:



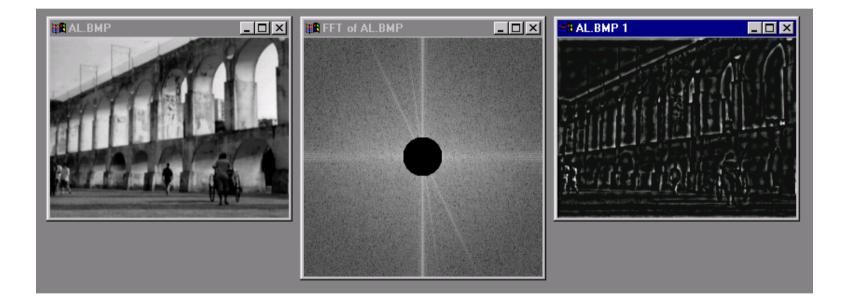
High-pass / band-pass:







Edges in images

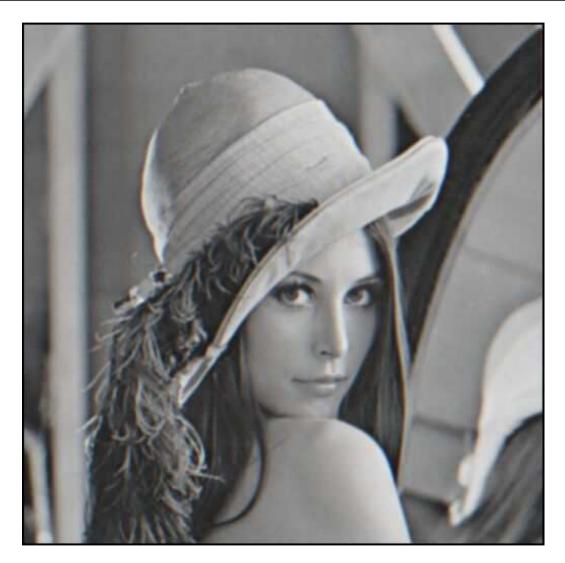


What does blurring take away?



original

What does blurring take away?



smoothed (5x5 Gaussian)

High-Pass filter



smoothed - original

Band-pass filtering

Gaussian Pyramid (low-pass images)

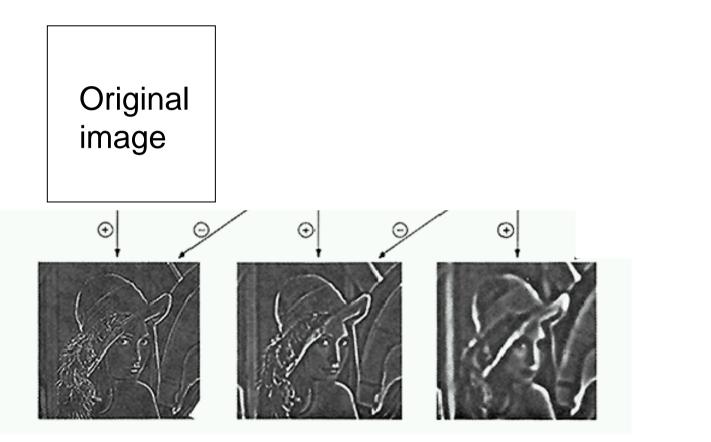






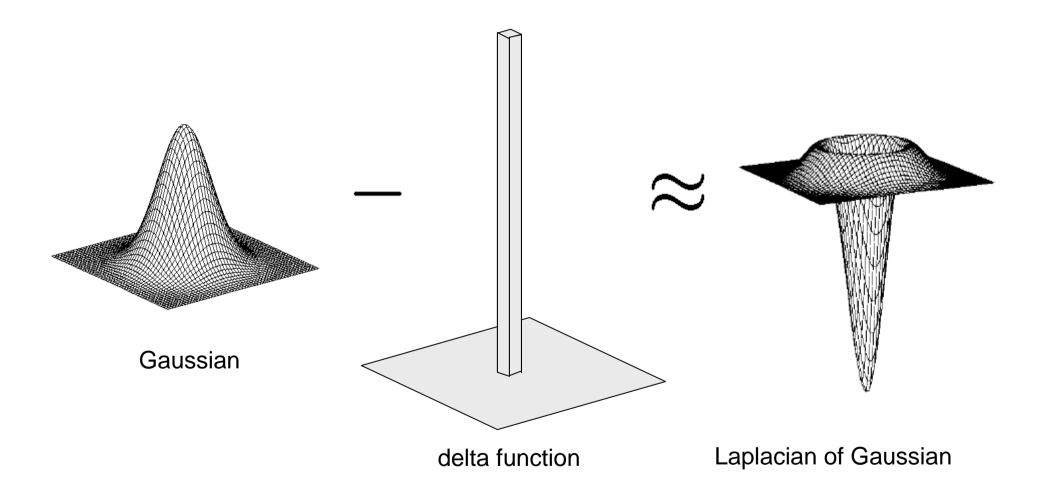


Laplacian Pyramid

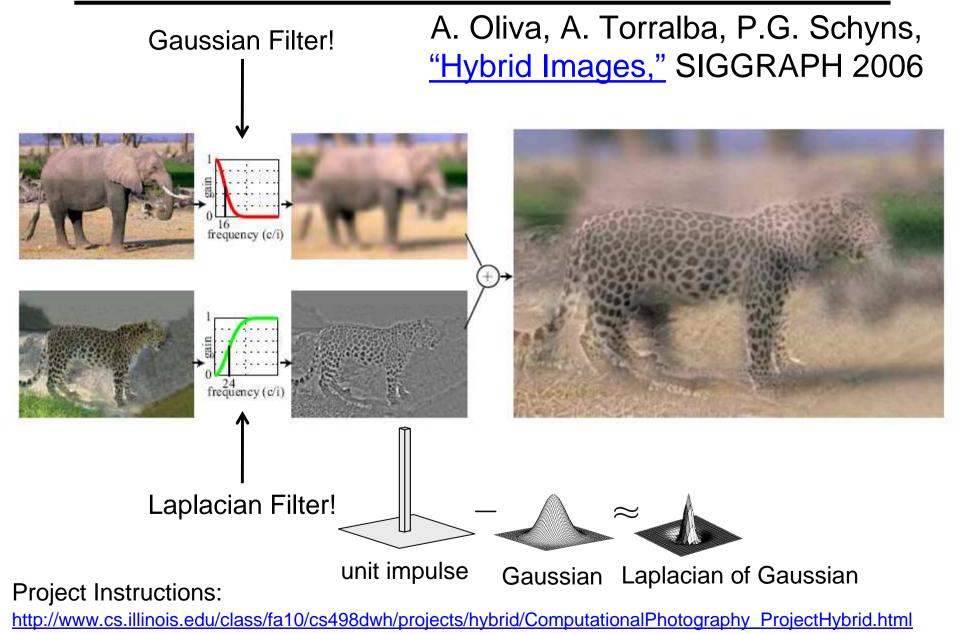


How can we reconstruct (collapse) this pyramid into the original image?

```
Why Laplacian?
```



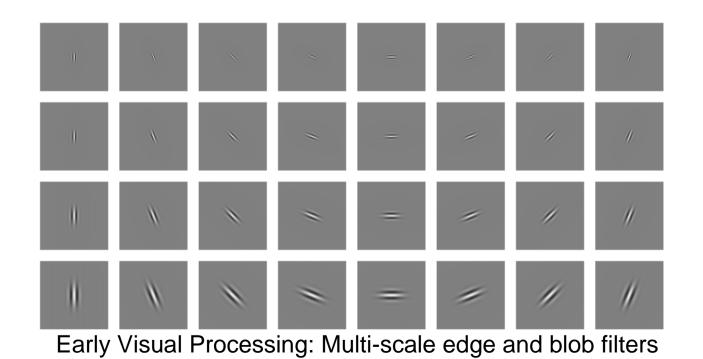
Project 1g: Hybrid Images



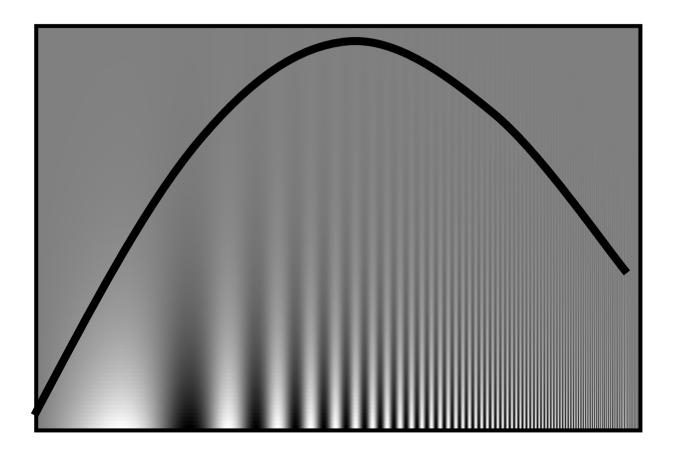
Clues from Human Perception

- Early processing in humans filters for various orientations and scales of frequency
- Perceptual cues in the mid frequencies dominate perception

When we see an image from far away, we are effectively subsampling it

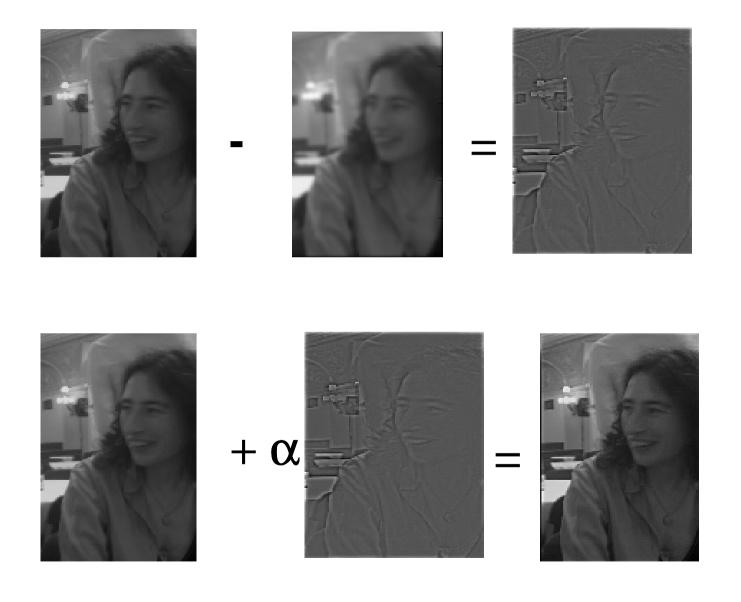


Frequency Domain and Perception

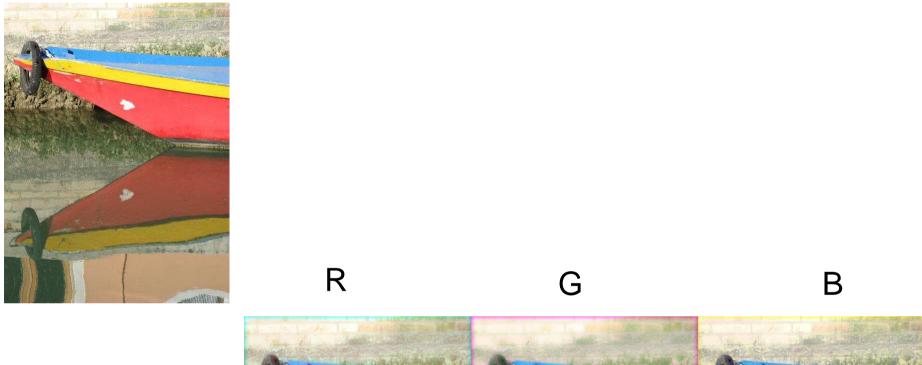


Campbell-Robson contrast sensitivity curve

Unsharp Masking

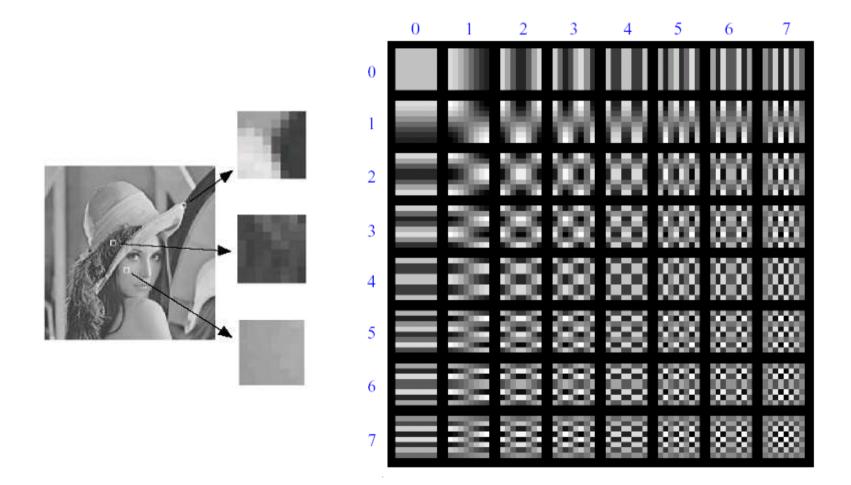


Freq. Perception Depends on Color





Lossy Image Compression (JPEG)



Block-based Discrete Cosine Transform (DCT)

Using DCT in JPEG

The first coefficient B(0,0) is the DC component, the average intensity

The top-left coeffs represent low frequencies, the bottom right – high frequencies

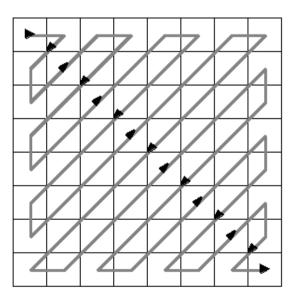


Image compression using DCT

Quantize

More coarsely for high frequencies (which also tend to have smaller • values)

40 51

61

55

56

62

77

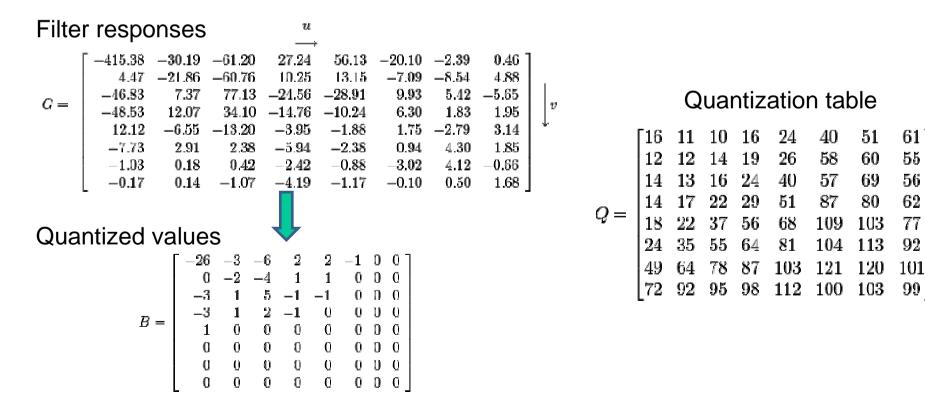
92

99

Many quantized high frequency values will be zero

Encode

Can decode with inverse dct



JPEG Compression Summary

Subsample color by factor of 2

• People have bad resolution for color

Split into blocks (8x8, typically), subtract 128

For each block

- a. Compute DCT coefficients for
- b. Coarsely quantize
 - Many high frequency components will become zero
- c. Encode (e.g., with Huffman coding)

Block size in JPEG

Block size

- small block
 - faster
 - correlation exists between neighboring pixels
- large block
 - better compression in smooth regions
- It's 8x8 in standard JPEG

JPEG compression comparison



89k



12k

Image gradient

The gradient of an image: $\nabla f = \left[\frac{\partial f}{\partial x}, \frac{\partial f}{\partial y}\right]$

The gradient points in the direction of most rapid change in intensity

$$\nabla f = \begin{bmatrix} \frac{\partial f}{\partial x}, 0 \end{bmatrix}$$

$$\nabla f = \begin{bmatrix} 0, \frac{\partial f}{\partial y} \end{bmatrix}$$

$$\nabla f = \begin{bmatrix} 0, \frac{\partial f}{\partial y} \end{bmatrix}$$

The gradient direction is given by:

$$\theta = \tan^{-1} \left(\frac{\partial f}{\partial y} / \frac{\partial f}{\partial x} \right)$$

• how does this relate to the direction of the edge?

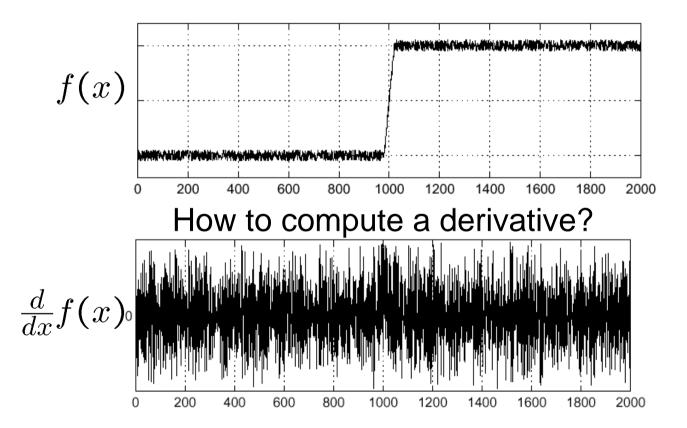
The *edge strength* is given by the gradient magnitude

$$\|\nabla f\| = \sqrt{\left(\frac{\partial f}{\partial x}\right)^2 + \left(\frac{\partial f}{\partial y}\right)^2}$$

Effects of noise

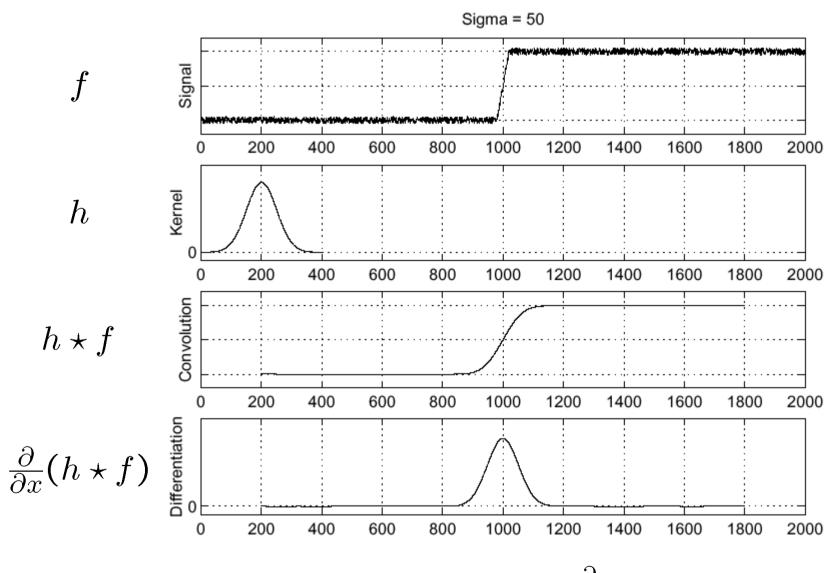
Consider a single row or column of the image

• Plotting intensity as a function of position gives a signal



Where is the edge?

Solution: smooth first

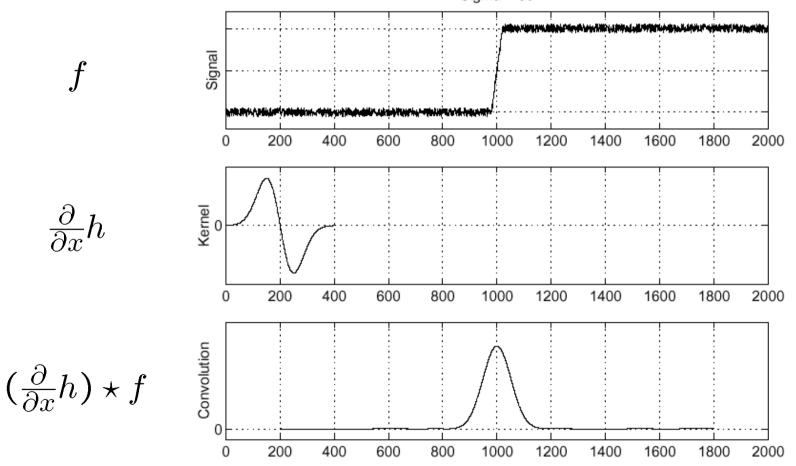


Where is the edge? Look for peaks in $\frac{\partial}{\partial x}(h \star f)$

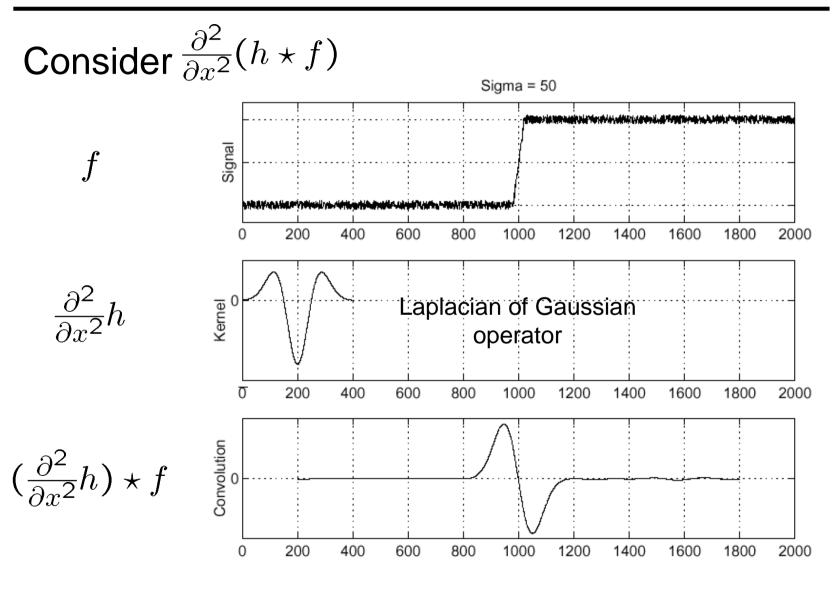
Derivative theorem of convolution

$$\frac{\partial}{\partial x}(h \star f) = (\frac{\partial}{\partial x}h) \star f$$

This saves us one operation:



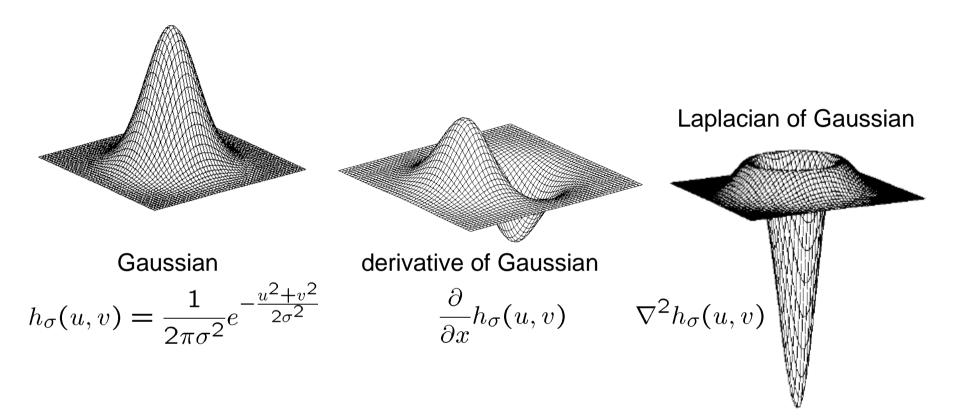
Laplacian of Gaussian



Where is the edge?

Zero-crossings of bottom graph

2D edge detection filters



 ∇^2 is the **Laplacian** operator:

$$\nabla^2 f = \frac{\partial^2 f}{\partial x^2} + \frac{\partial^2 f}{\partial y^2}$$

Try this in MATLAB

```
g = fspecial('gaussian',15,2);
imagesc(g); colormap(gray);
surfl(g)
gclown = conv2(clown,g,'same');
imagesc(conv2(clown, [-1 1], 'same'));
imagesc(conv2(gclown,[-1 1],'same'));
dx = conv2(g, [-1 \ 1], 'same');
imagesc(conv2(clown,dx,'same'));
lg = fspecial('log', 15, 2);
lclown = conv2(clown,lg,'same');
imagesc(lclown)
imagesc(clown + .2*lclown)
```