

# BBM 413

## Fundamentals of Image Processing

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### Frequency Domain Techniques – Part I

## Review - Point Operations

- Smallest possible neighborhood is of size 1x1
- Process each point independently of the others
- Output image  $g$  depends only on the value of  $f$  at a single point  $(x,y)$
- Transformation function  $T$  remaps the sample's value:

$$s = T(r)$$

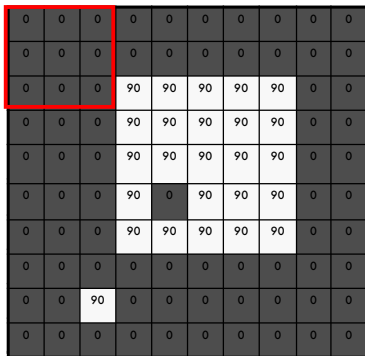
where

- $r$  is the value at the point in question
- $s$  is the new value in the processed result
- $T$  is a *intensity transformation function*

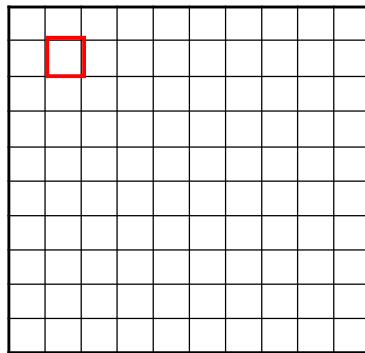
## Review – Spatial Filtering

$$g[\cdot, \cdot] = \frac{1}{9} \begin{bmatrix} 1 & 1 & 1 \\ 1 & 1 & 1 \\ 1 & 1 & 1 \end{bmatrix}$$

$f[.,.]$



$h[.,.]$



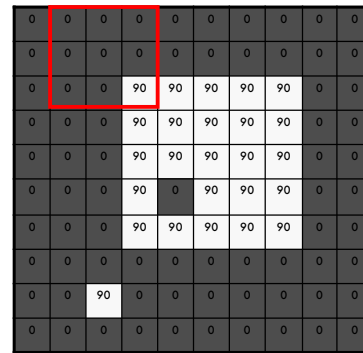
$$h[m,n] = \sum_{k,l} g[k,l] f[m+k,n+l]$$

Slide credit: S. Seitz

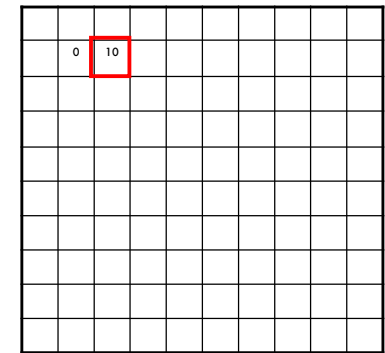
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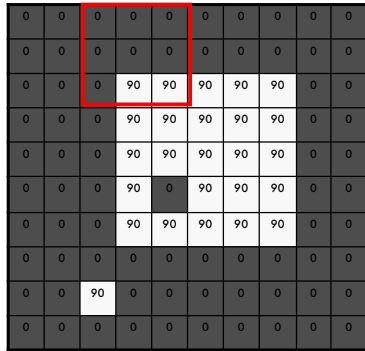
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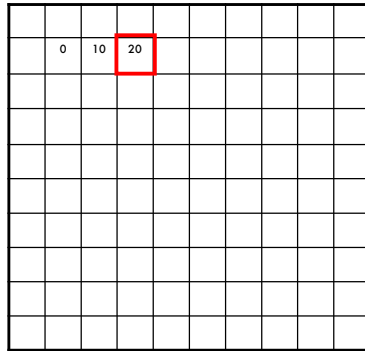
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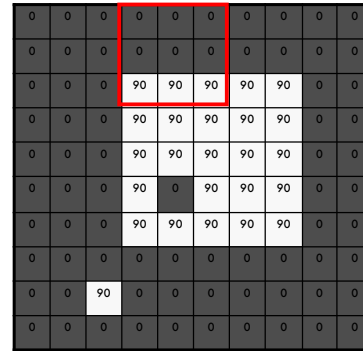
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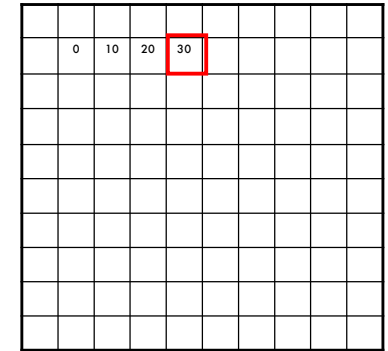
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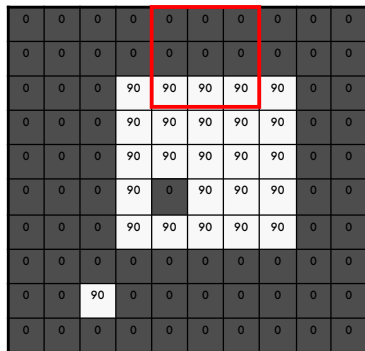
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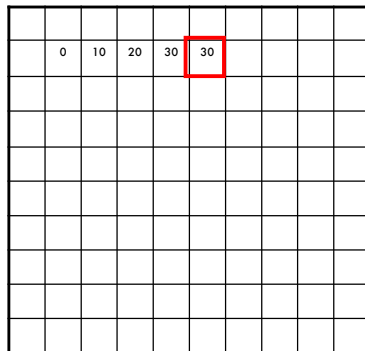
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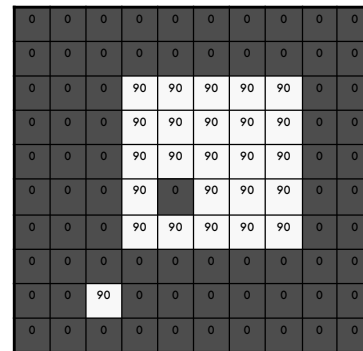
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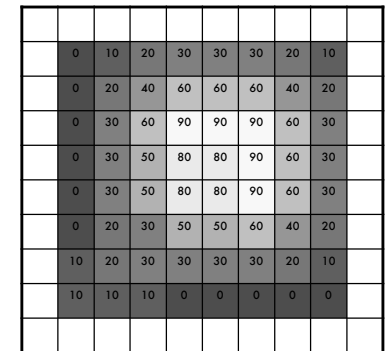
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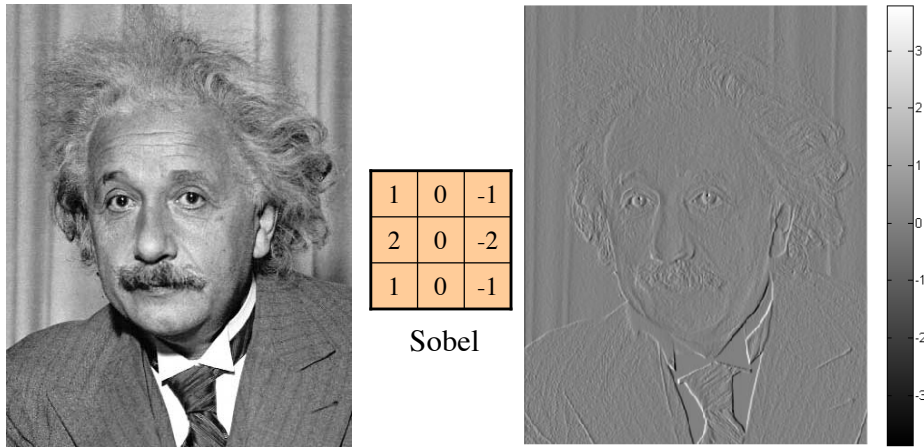
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Slide credit: S. Seitz

## Review – Spatial Filtering



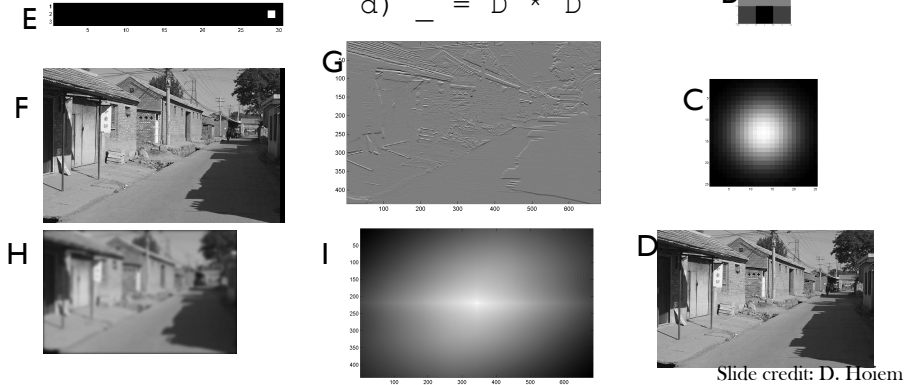
Slide credit: J. Hays

## Review – Spatial Filtering

Fill in the blanks:

- a)  $\_ = D * B$   
 b)  $\bar{A} = \_ * \_$   
 c)  $F = D * \_$   
 d)  $\_ = D * D$

Filtering Operator

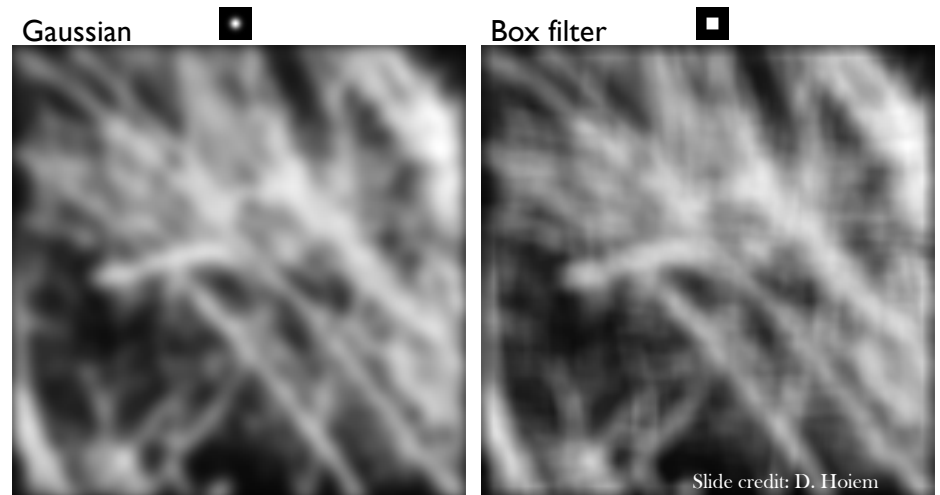


Slide credit: D. Hoem

## Today

- Frequency domain techniques
- Images in terms of frequency
- Fourier Series
- Convolution Theorem

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



Slide credit: D. Hoem

**Why does a lower resolution image still make sense to us? What do we lose?**



Image: <http://www.flickr.com/photos/igorms/136916757/>

Slide credit: D. Hoiem

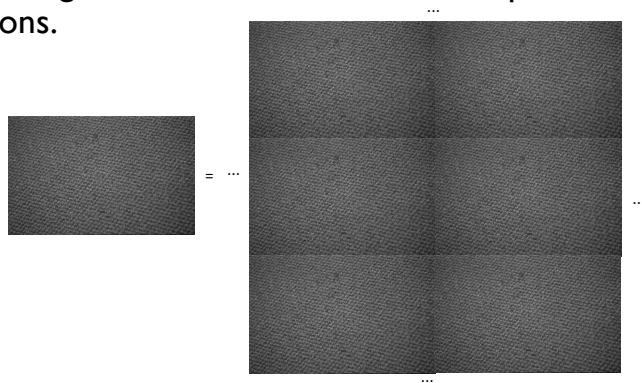
**How is it that a 4MP image can be compressed to a few hundred KB without a noticeable change?**



Slide credit: J. Hays

## Answer to these questions?

- Thinking images in terms of frequency.
- Treat images as infinite-size, continuous periodic functions.



## Jean Baptiste Joseph Fourier (1768-1830)

had crazy idea (1807):

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



Slide credit: A. Efros

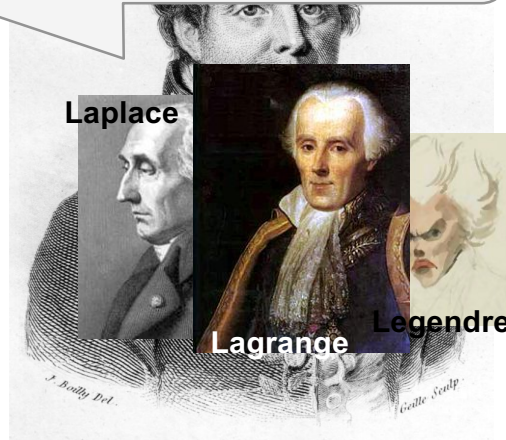
## Jean Baptiste Joseph Fourier (1768-1830)

had crazy idea

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

...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.

- Don't believe it?
  - Neither did Lagrange, Laplace, Poisson and other big wigs
  - Not translated into English until 1878!



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- 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



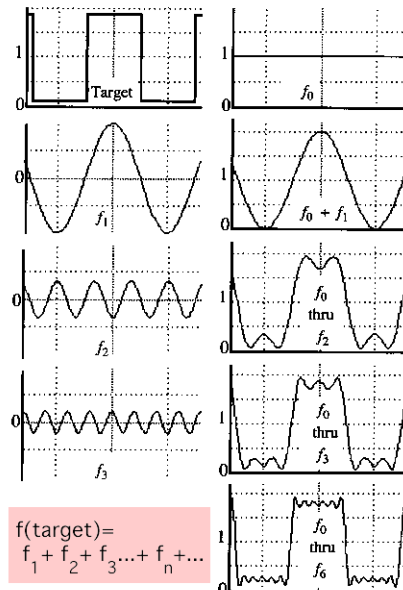
Slide credit: A. Efros

## A sum of sines

Our building block:

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

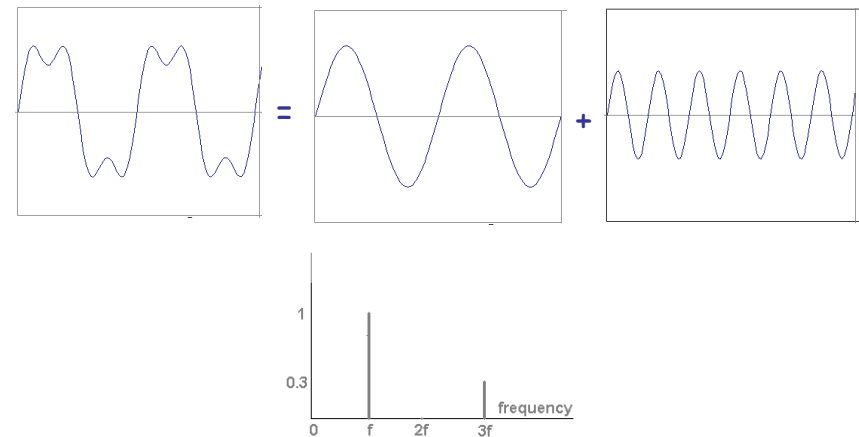
Add enough of them to get any signal  $f(x)$  you want!



Slide credit: A. Efros

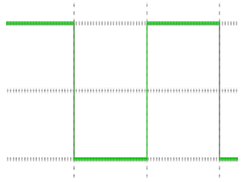
## Frequency Spectra

- example:  $g(t) = \sin(2\pi f t) + (1/3)\sin(2\pi(3f) t)$



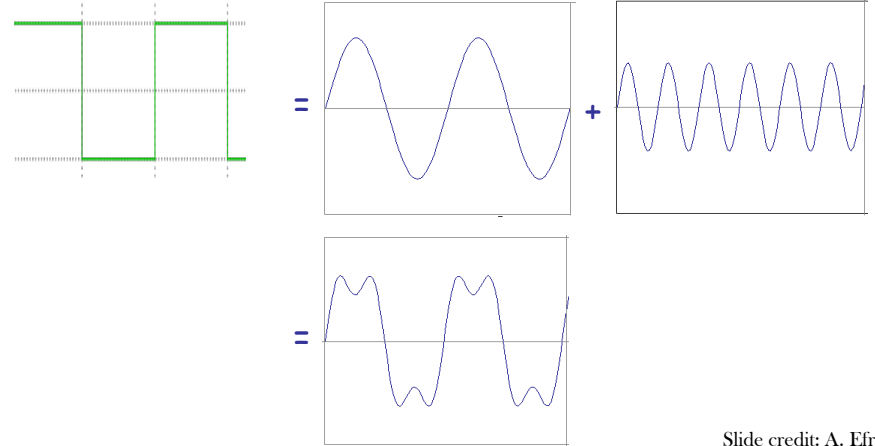
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# Frequency Spectra



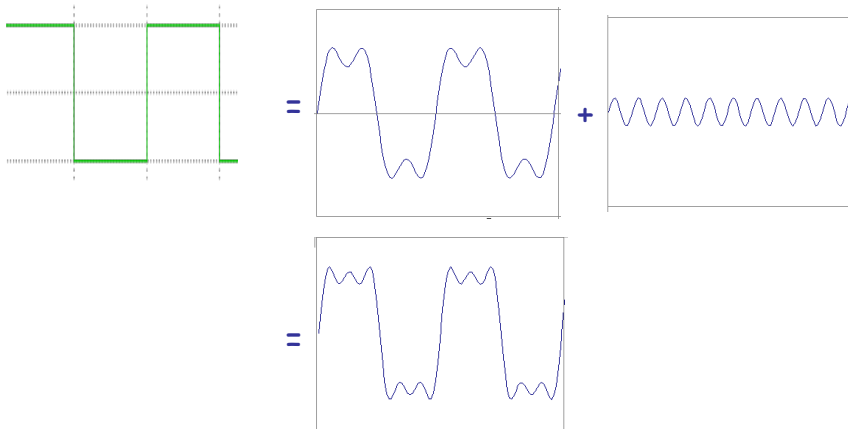
Slide credit: A. Efros

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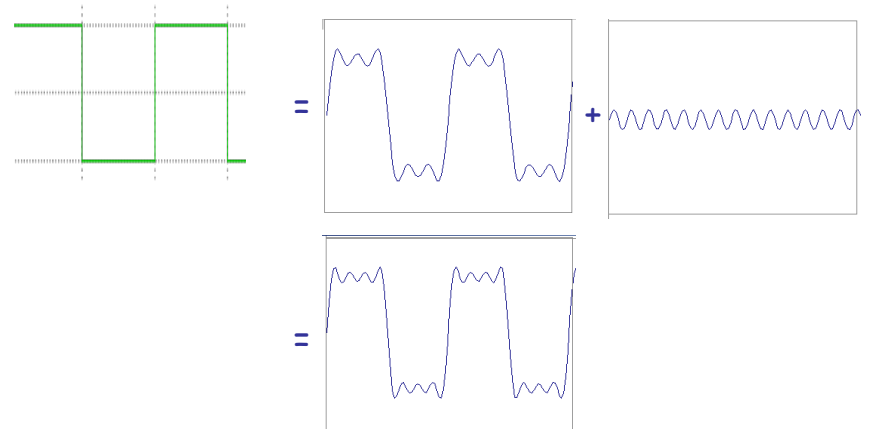
Slide credit: A. Efros

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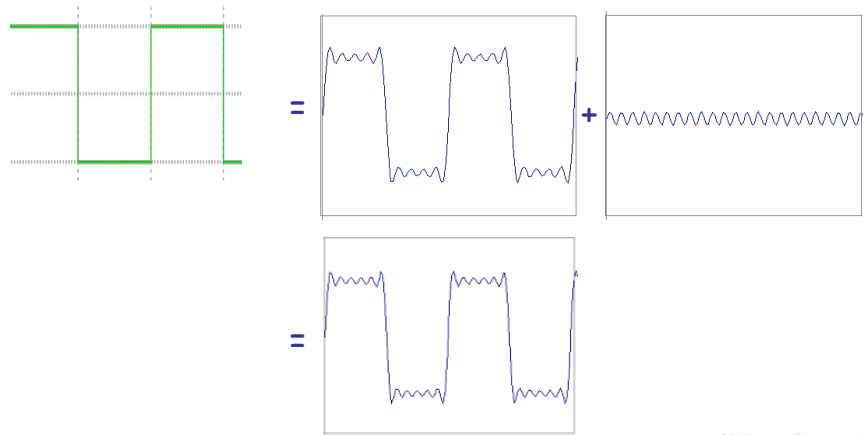
Slide credit: A. Efros

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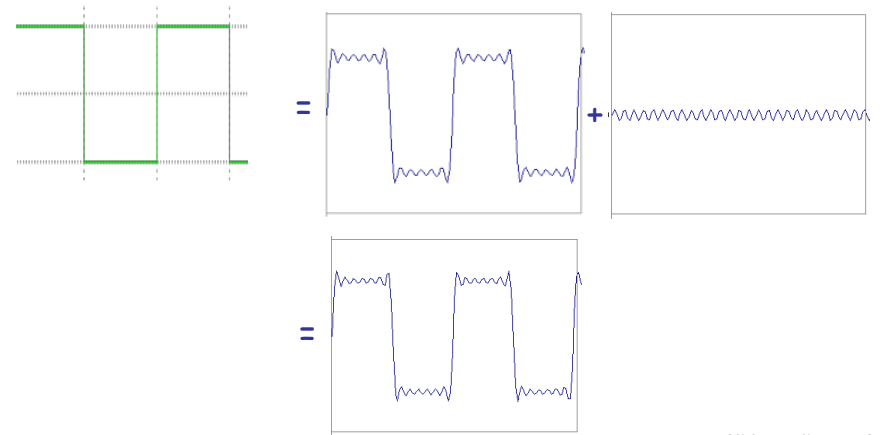
Slide credit: A. Efros

## Frequency Spectra



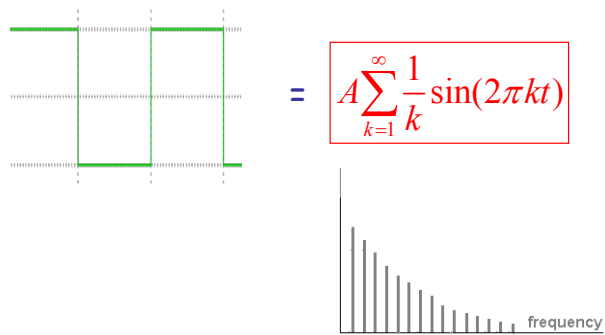
Slide credit: A. Efros

## Frequency Spectra



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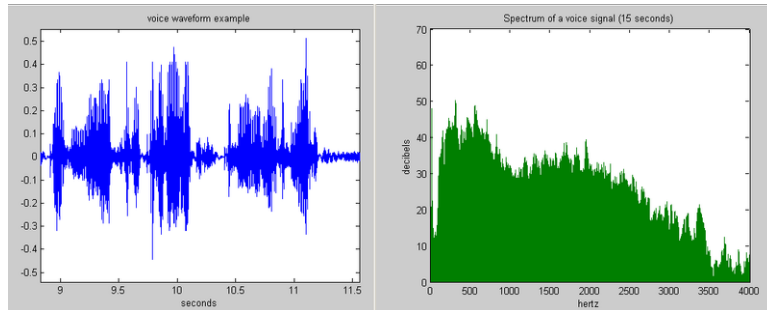
## Frequency Spectra



Image credit: Lucas V. Barbosa

## Example: Music

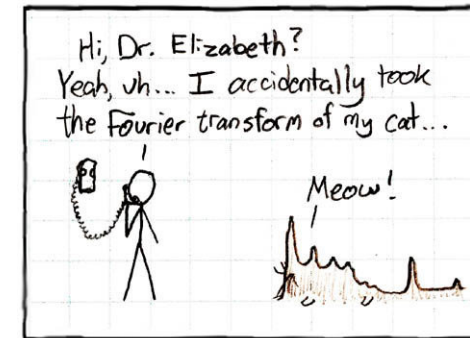
- We think of music in terms of frequencies at different magnitudes.



Slide credit: D. Hoem

## Other signals

- We can also think of all kinds of other signals the same way

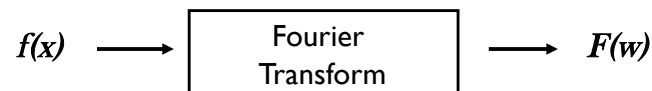


xkcd.com

Slide credit: J. Hays

## Fourier Transform

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



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

- How can  $F$  hold both? Complex number trick!

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

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

We can always go back:



Slide credit: A. Efros

## Fourier Transform

- Fourier transform stores the magnitude and phase at each frequency
  - Magnitude encodes how much signal there is at a particular frequency
  - Phase encodes spatial information (indirectly)
  - For mathematical convenience, this is often notated in terms of real and complex numbers

$$\text{Amplitude: } A = \pm \sqrt{R(\omega)^2 + I(\omega)^2} \quad \text{Phase: } \phi = \tan^{-1} \frac{I(\omega)}{R(\omega)}$$



## Discrete Fourier transform

- Forward transform

$$F(u, v) = \frac{1}{MN} \sum_{x=0}^{M-1} \sum_{y=0}^{N-1} f(x, y) e^{-j2\pi(ux/M + vy/N)}$$

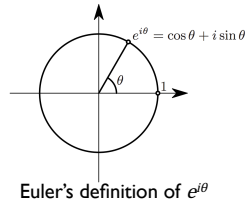
for  $u = 0, 1, 2, \dots, M-1, v = 0, 1, 2, \dots, N-1$

- Inverse transform

$$f(x, y) = \sum_{u=0}^{M-1} \sum_{v=0}^{N-1} F(u, v) e^{j2\pi(ux/M + vy/N)}$$

for  $x = 0, 1, 2, \dots, M-1, y = 0, 1, 2, \dots, N-1$

$u, v$  : the transform or frequency variables  
 $x, y$  : the spatial or image variables



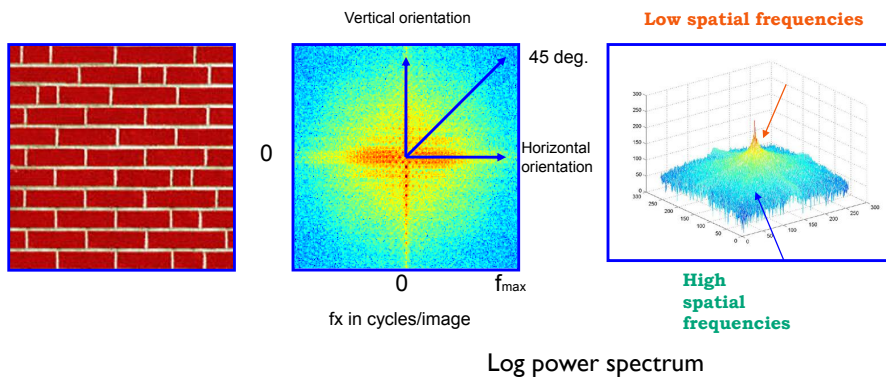
Slide credit: B. Freeman and A. Torralba

## The Fourier Transform

- Represent function on a new basis
  - Think of functions as vectors, with many components
  - We now apply a linear transformation to transform the basis
    - dot product with each basis element
- In the expression,  $u$  and  $v$  select the basis element, so a function of  $x$  and  $y$  becomes a function of  $u$  and  $v$
- basis elements have the form  $e^{-i2\pi(ux+vy)}$

Slide credit: S. Thrum

## How to interpret 2D Fourier Spectrum



Slide credit: B. Freeman and A. Torralba

Fourier basis element  
 $e^{-i2\pi(ux+vy)}$

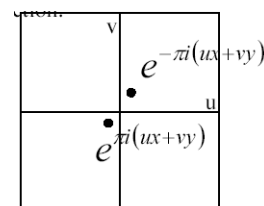
example, real part

$F^{u,v}(x, y)$

$F^{u,v}(x, y) = \text{const. for } (ux+vy) = \text{const.}$

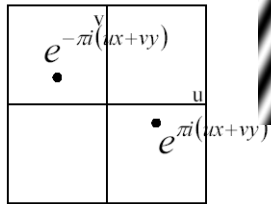
Vector  $(u, v)$

- Magnitude gives frequency
- Direction gives orientation.



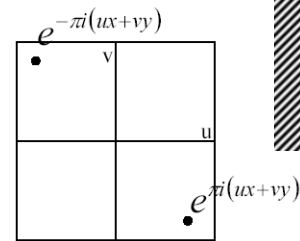
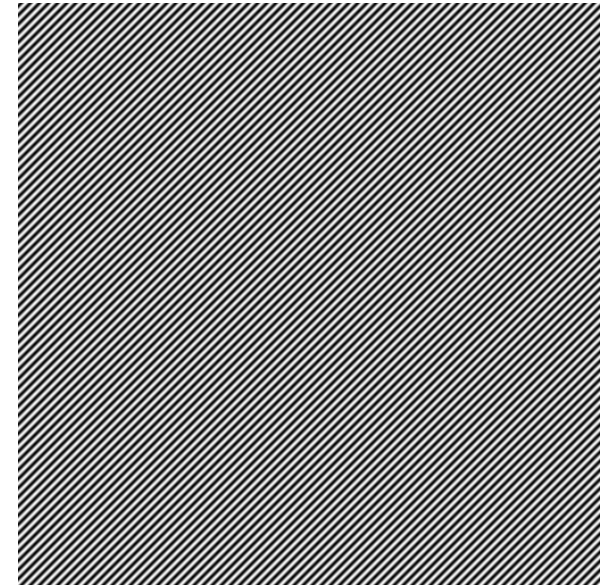
Slide credit: S. Thrum

Here  $u$  and  $v$  are larger than in the previous slide.



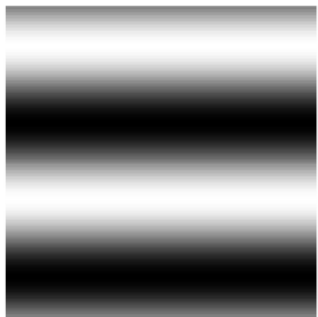
Slide credit: S. Thrun

And larger still...



Slide credit: S. Thrun

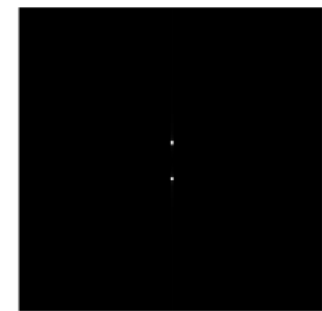
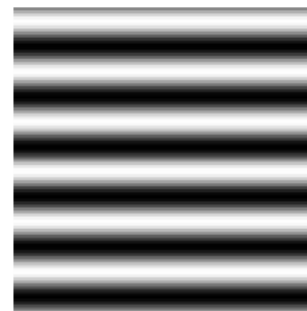
## 2D FFT



Sinusoid with frequency = 1 and its FFT

Slide credit: M. H. Yang

## 2D FFT



Sinusoid with frequency = 3 and its FFT

Slide credit: M. H. Yang

## 2D FFT



Sinusoid with frequency = 5 and its FFT

Slide credit: M. H. Yang

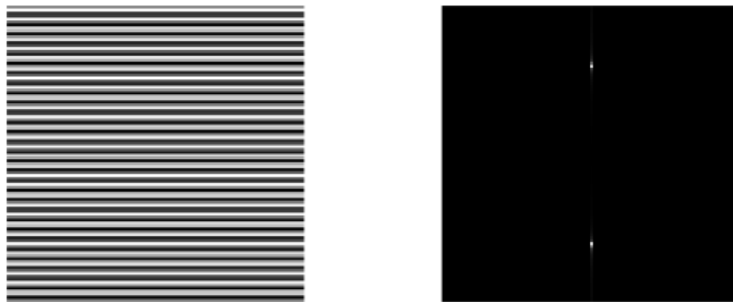
## 2D FFT



Sinusoid with frequency = 10 and its FFT

Slide credit: M. H. Yang

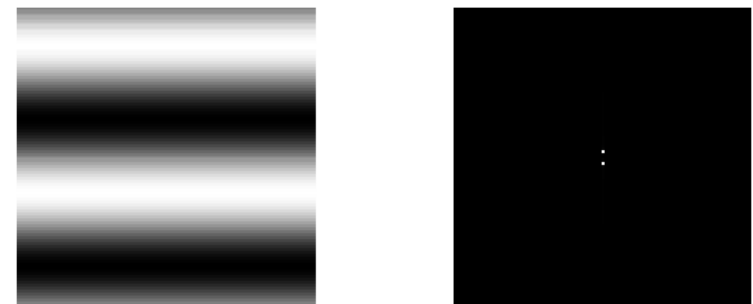
## 2D FFT



Sinusoid with frequency = 15 and its FFT

Slide credit: M. H. Yang

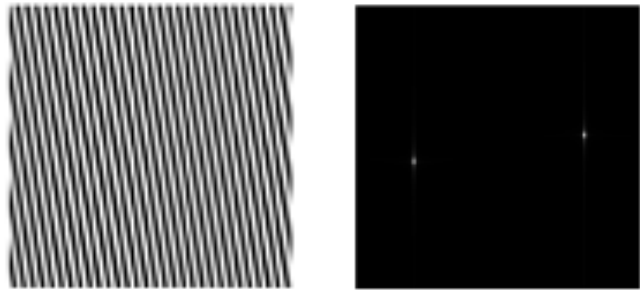
## 2D FFT



Sinusoid with varying frequency and their FFT

Slide credit: M. H. Yang

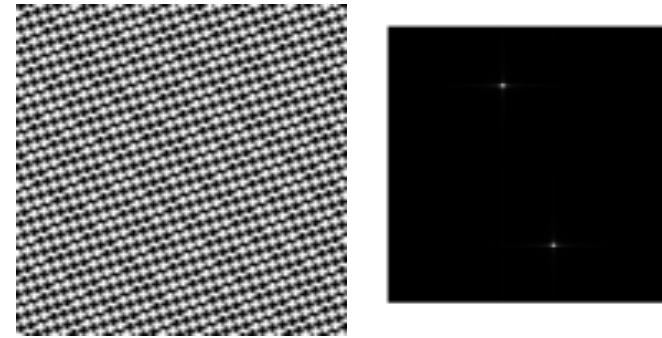
## Rotation



Sinusoid rotated at 30 degrees and its FFT

Slide credit: M. H. Yang

## 2D FFT

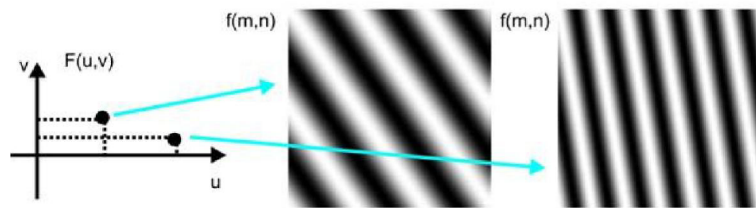


Sinusoid rotated at 60 degrees and its FFT

Slide credit: M. H. Yang

## 2D FFT

$$F(u, v) = \frac{1}{MN} \cdot \sum_{x=0}^{M-1} \sum_{y=0}^{N-1} f(x, y) e^{-i2\pi(xu/M + yv/N)}$$

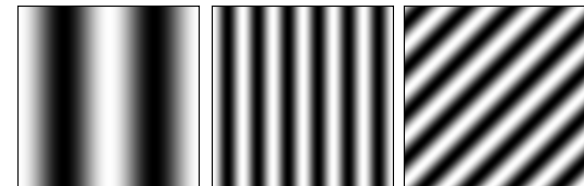


Convolution masks for different frequencies

Slide credit: M. H. Yang

## Fourier analysis in images

Intensity Image



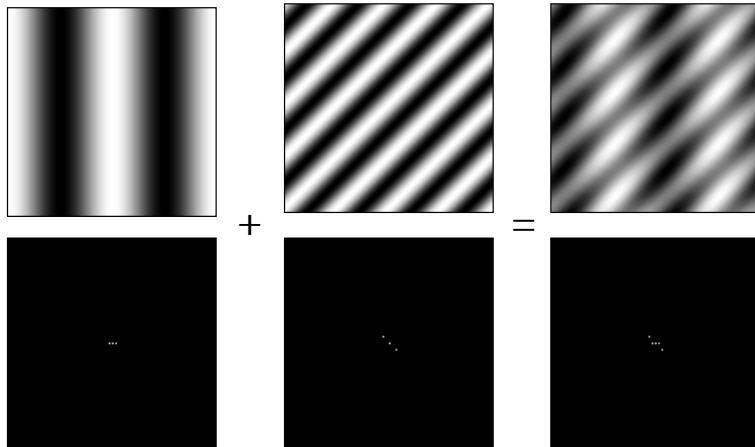
Fourier Image



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

Slide credit: A. Efros

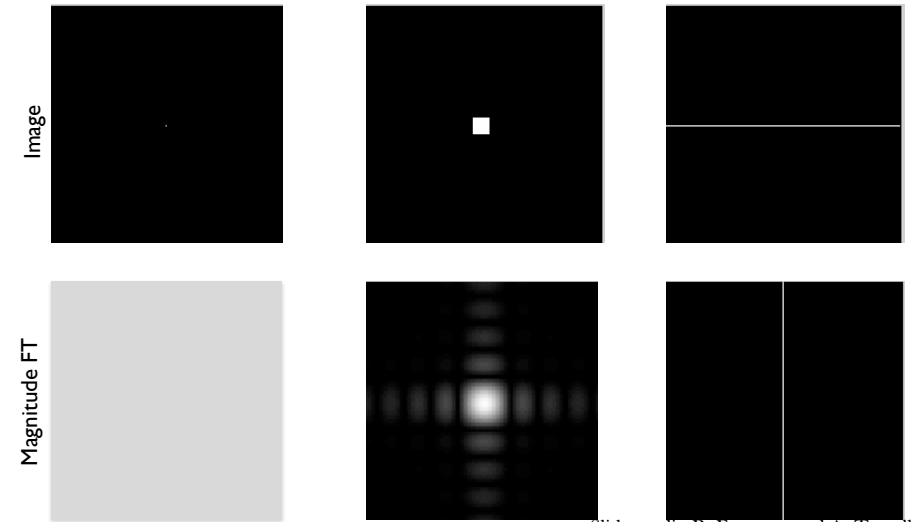
## Signals can be composed



Slide credit: A. Efros

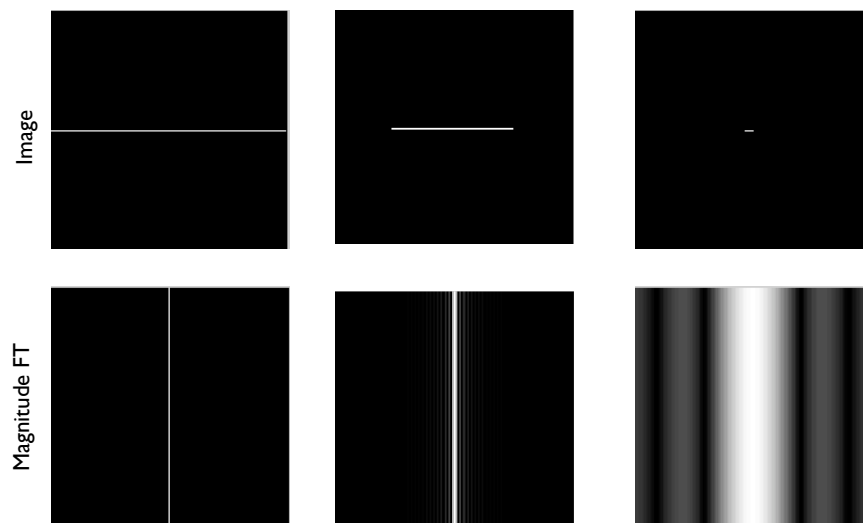
<http://sharp.bu.edu/~slehar/fourier/fourier.html#filtering>  
 More: <http://www.cs.unm.edu/~brayer/vision/fourier.html>

## Some important Fourier Transforms



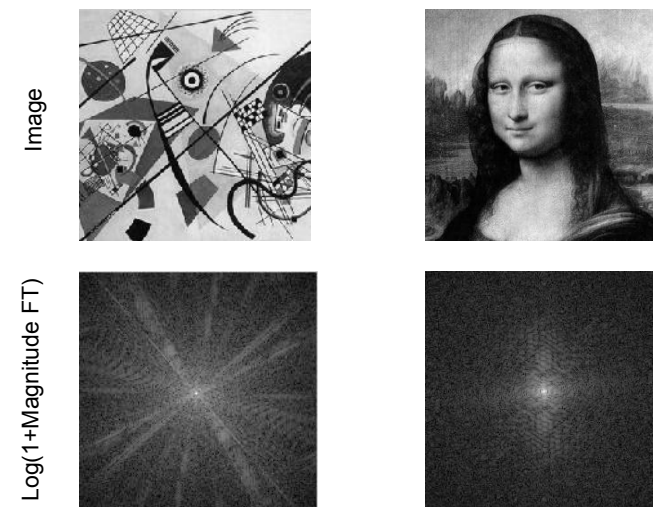
Slide credit: B. Freeman and A. Torralba

## Some important Fourier Transforms



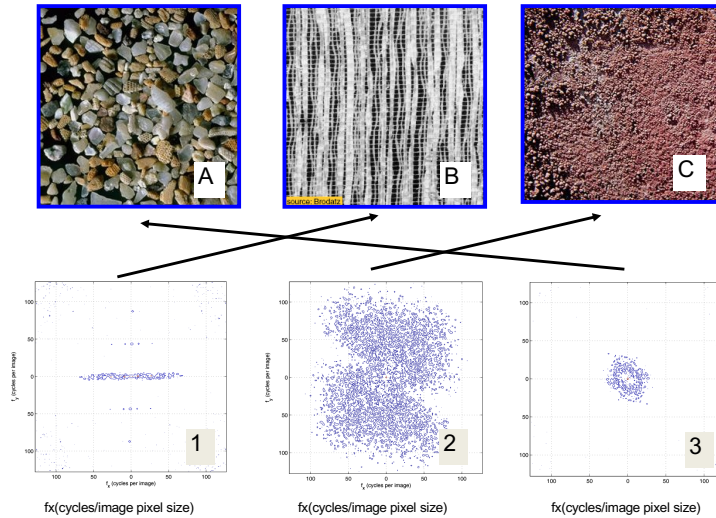
Slide credit: B. Freeman and A. Torralba

## The Fourier Transform of some well-known images



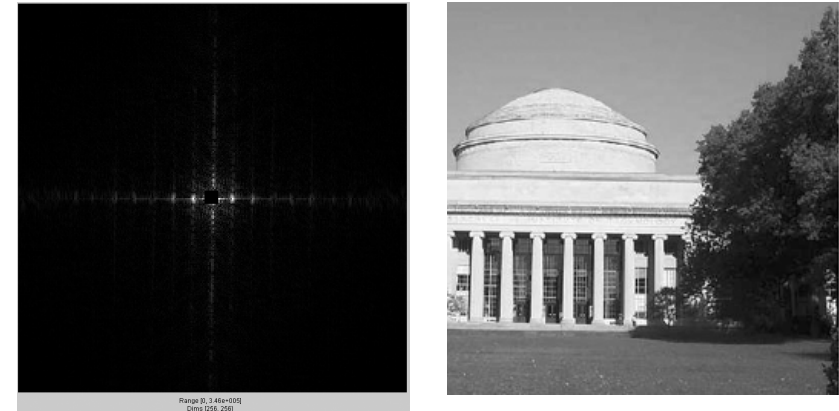
Slide credit: B. Freeman and A. Torralba

## Fourier Amplitude Spectrum



Slide credit: B. Freeman and A. Torralba

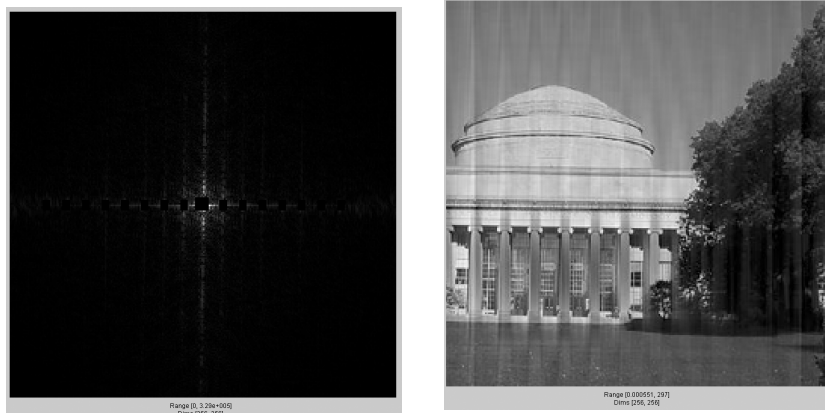
## Fourier transform magnitude



What in the image causes the dots?

Slide credit: B. Freeman and A. Torralba

## Masking out the fundamental and harmonics from periodic pillars



Slide credit: B. Freeman and A. Torralba

## The Convolution Theorem

- 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!

Slide credit: A. Efros

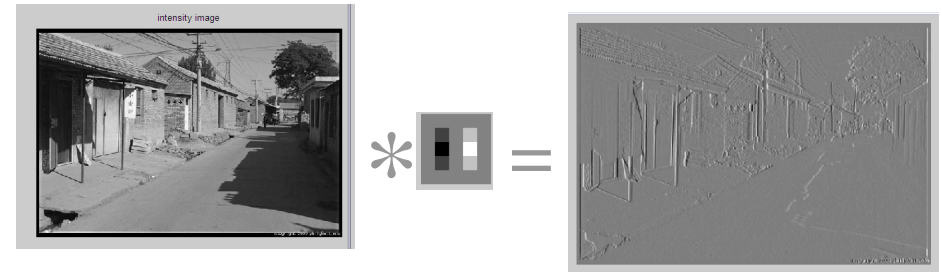
## Properties of Fourier Transforms

- Linearity  $\mathcal{F}[ax(t) + by(t)] = a\mathcal{F}[x(t)] + b\mathcal{F}[y(t)]$
- Fourier transform of a real signal is symmetric about the origin
- The energy of the signal is the same as the energy of its Fourier transform

Slide credit: J. Hays

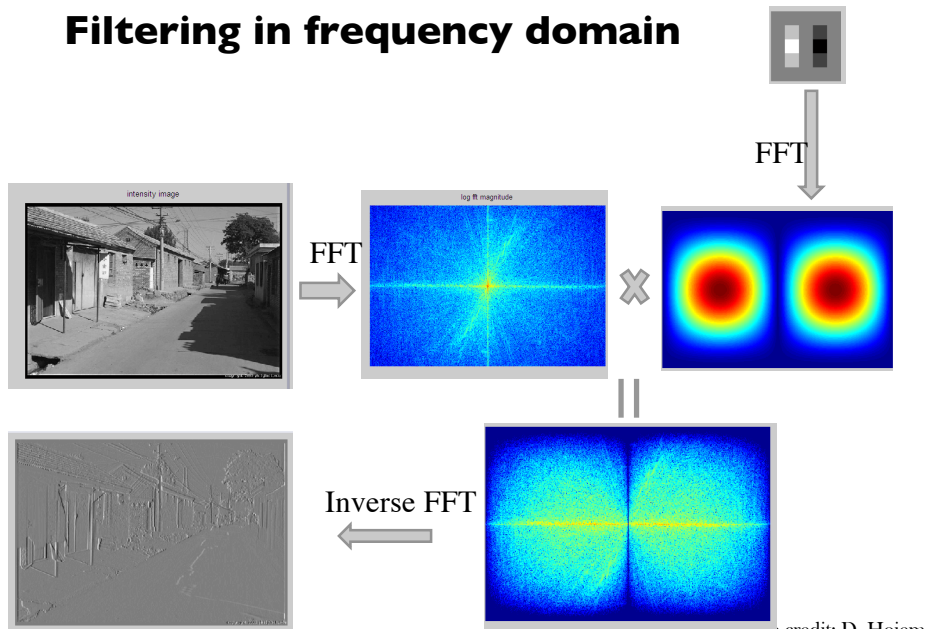
## Filtering in spatial domain

1	0	-1
2	0	-2
1	0	-1



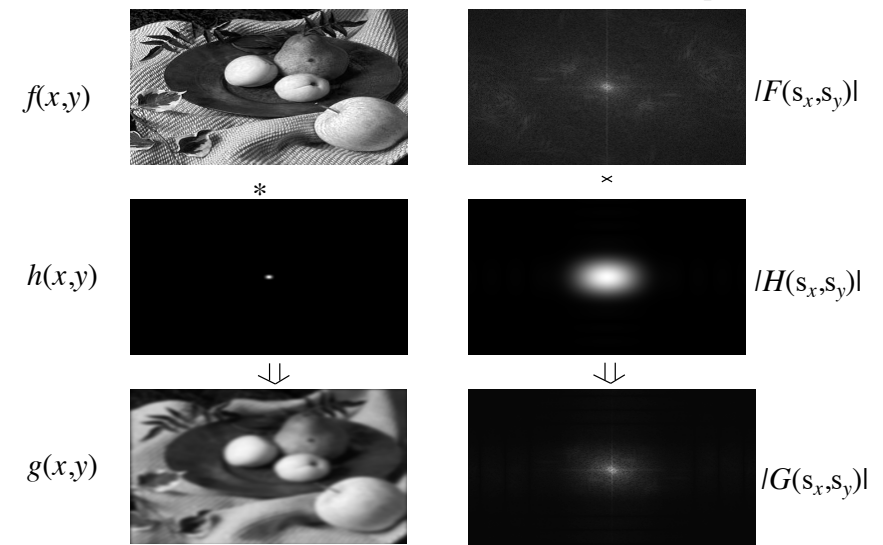
Slide credit: D. Hoiem

## Filtering in frequency domain



Slide credit: D. Hoiem

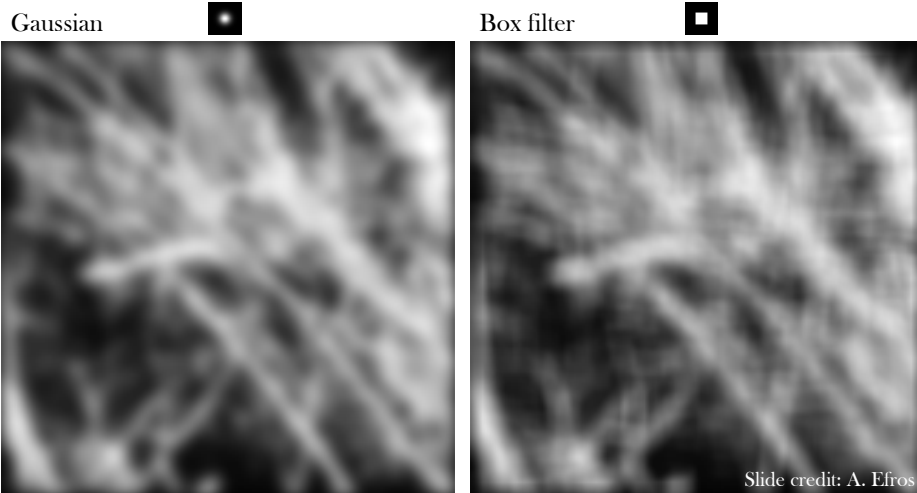
## 2D convolution theorem example



Slide credit: A. Efros

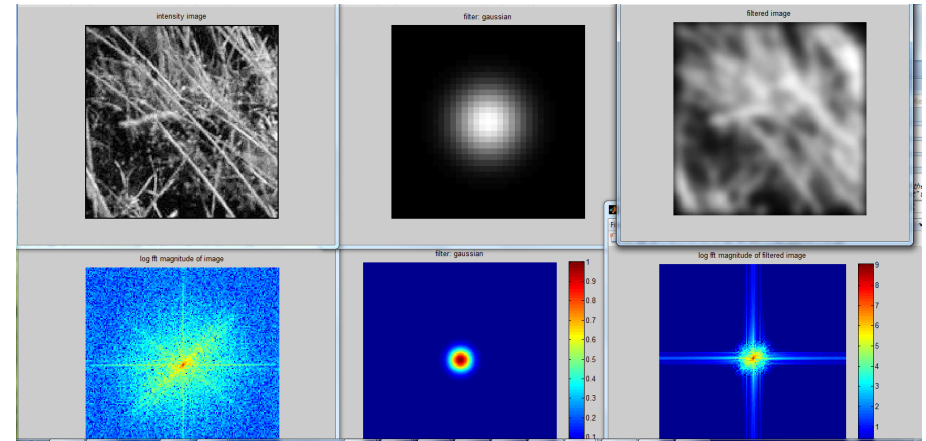
## Filtering

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



## Filtering

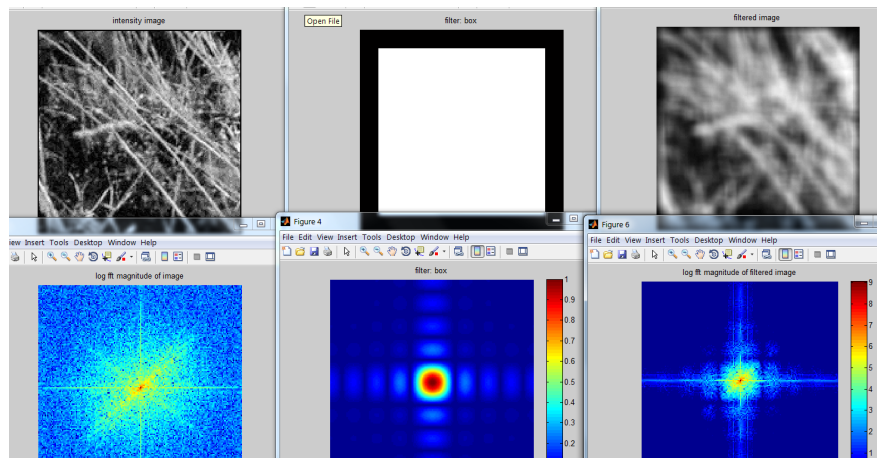
Gaussian



Slide credit: A. Efros

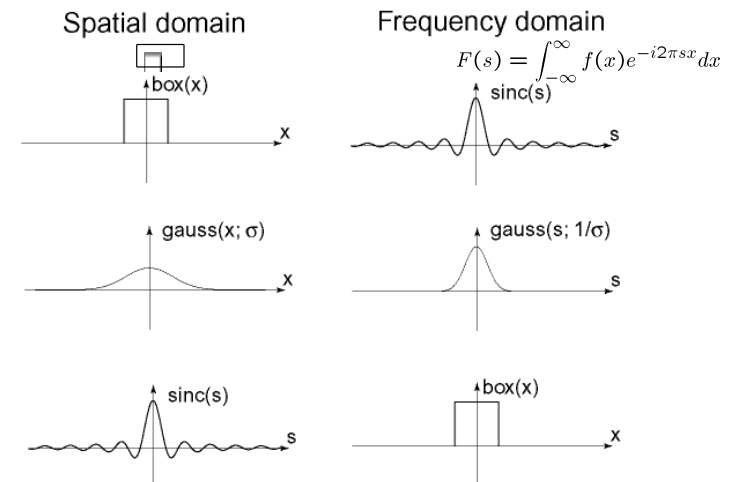
## Filtering

Box Filter



Slide credit: A. Efros

## Fourier Transform pairs

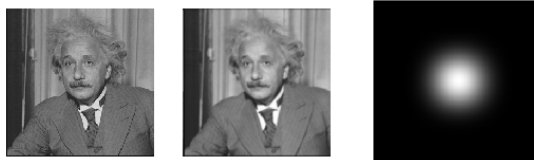


Slide credit: A. Efros



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

low-pass:

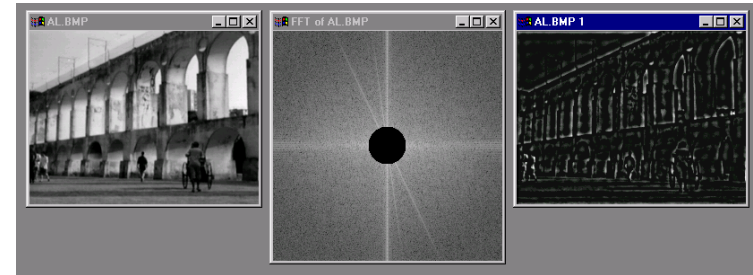


High-pass / band-pass:



Slide credit: A. Efros

## Edges in images



Slide credit: A. Efros

## FFT in Matlab

### • Filtering with fft

```
im = ... % "im" should be a gray-scale floating point image
[imh, imw] = size(im);
fftsize = 1024; % should be order of 2 (for speed) and include padding
im_fft = fft2(im, fftsize, fftsize); % 1) fft im with padding
hs = 50; % filter half-size
fil = fspecial('gaussian', hs*2+1, 10);
fil_fft = fft2(fil, fftsize, fftsize); % 2) fft fil, pad to same size as image
im_fil_fft = im_fft .* fil_fft; % 3) multiply fft images
im_fil = ifft2(im_fil_fft); % 4) inverse fft2
im_fil = im_fil(1+hs:size(im,1)+hs, 1+hs:size(im, 2)+hs); % 5) remove padding
```

### • Displaying with fft

```
figure(1), imagesc(log(abs(fftshift(im_fft)))), axis image, colormap
jet
```

Slide credit: D. Hoiem

## Phase and Magnitude

### • Curious fact

- all natural images have about the same magnitude transform
- hence, phase seems to matter, but magnitude largely doesn't

### • Demonstration

- Take two pictures, swap the phase transforms, compute the inverse - what does the result look like?



Image with cheetah phase  
(and zebra magnitude)

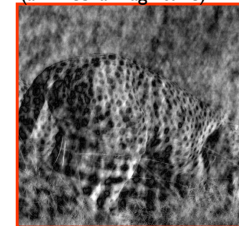
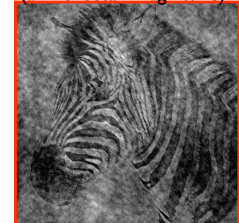


Image with zebra phase  
(and cheetah magnitude)

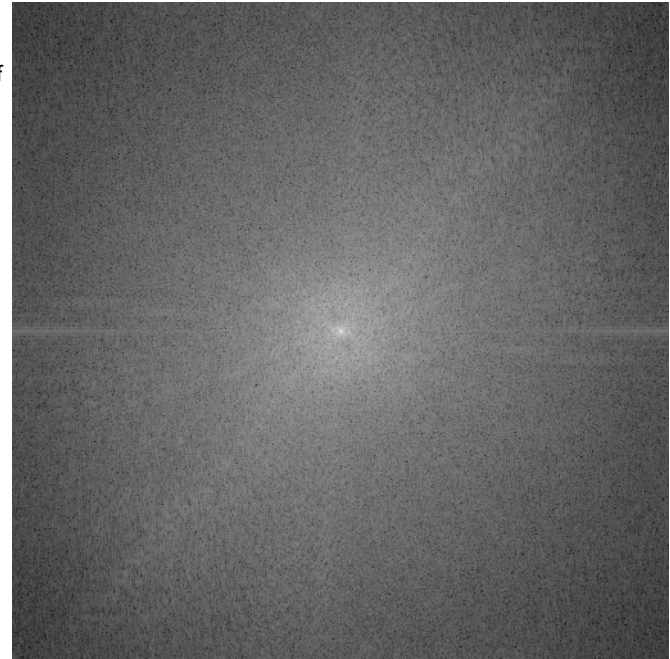


Slide credit: B. Freeman and A. Torralba



Slide credit: B. Freeman and A. Torralba

This is the  
magnitude  
transform of  
the cheetah  
picture

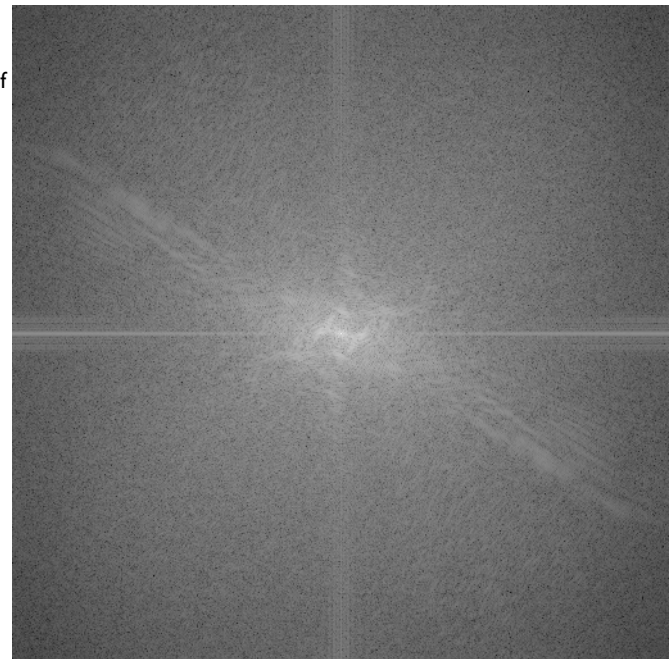


Slide credit: B. Freeman and A. Torralba



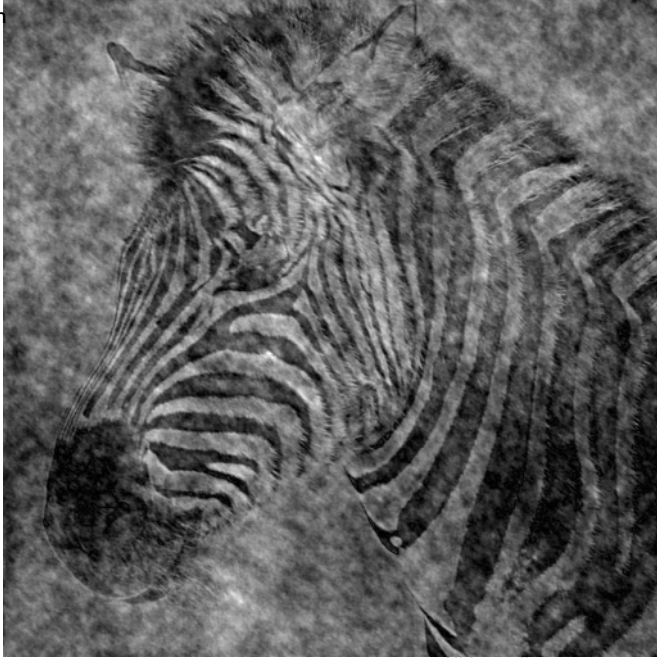
Slide credit: B. Freeman and A. Torralba

This is the  
magnitude  
transform of  
the zebra  
picture



Slide credit: B. Freeman and A. Torralba

Reconstruction  
with zebra  
phase, cheetah  
magnitude



Slide credit: B. Freeman and A. Torralba

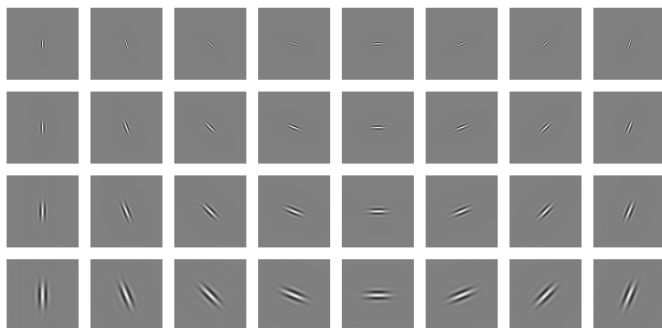
Reconstruction  
with cheetah  
phase, zebra  
magnitude



Slide credit: B. Freeman and A. Torralba

## Clues from Human Perception

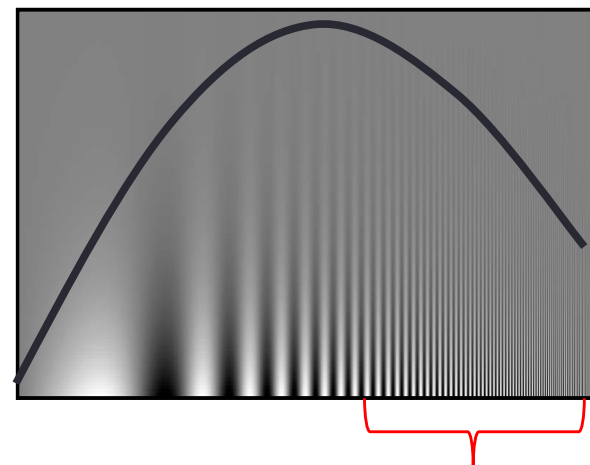
- Early processing in humans filters for various orientations and scales of frequency
- Perceptual cues in the mid-high frequencies dominate perception
- When we see an image from far away, we are effectively subsampling it



Early Visual Processing: Multi-scale edge and blob filters

Slide credit: J. Hays

## Campbell-Robson contrast sensitivity curve

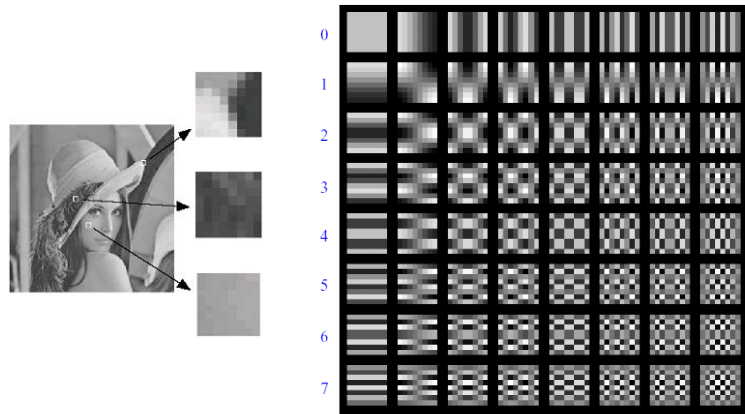


*The higher the frequency the less sensitive human visual system is...*

Slide credit: J. Hays

## Lossy Image Compression (JPEG)

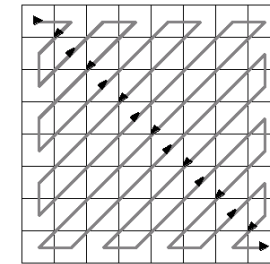
$$X_{k_1, k_2} = \sum_{n_1=0}^{N_1-1} \sum_{n_2=0}^{N_2-1} x_{n_1, n_2} \cos \left[ \frac{\pi}{N_1} \left( n_1 + \frac{1}{2} \right) k_1 \right] \cos \left[ \frac{\pi}{N_2} \left( n_2 + \frac{1}{2} \right) k_2 \right]$$



Block-based Discrete Cosine Transform (DCT) on 8x8 Slide credit: A. Bobick

## Using DCT in JPEG

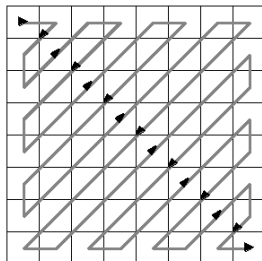
- The first coefficient  $B(0,0)$  is the DC component, the average intensity
- The top-left coeffs represent low frequencies, the bottom right – high frequencies



Slide credit: A. Bobick

## Image compression using DCT

- DCT enables image compression by concentrating most image information in the low frequencies
- Loose unimportant image info (high frequencies) by cutting  $B(u,v)$  at bottom right
- The decoder computes the inverse DCT – IDCT



Slide credit: A. Bobick

## JPEG compression comparison



89k

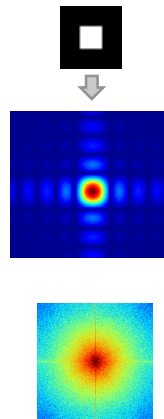


12k

Slide credit: A. Bobick

## Things to Remember

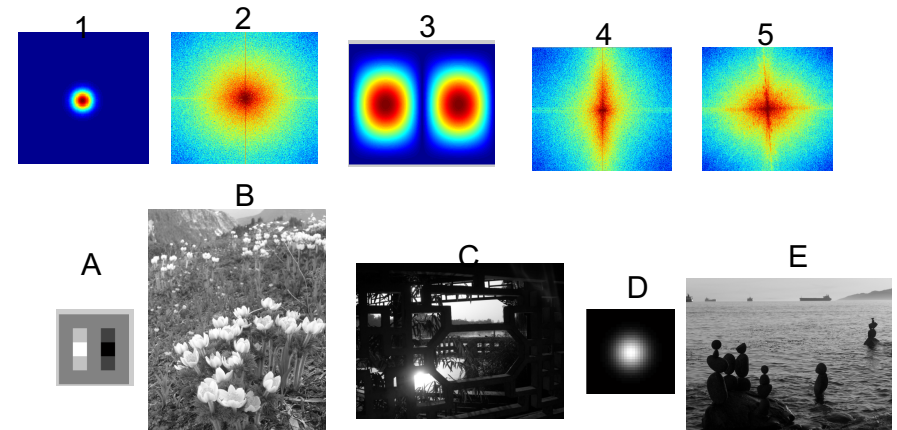
- Sometimes it makes sense to think of images and filtering in the frequency domain
  - Fourier analysis
- Can be faster to filter using FFT for large images ( $N \log N$  vs.  $N^2$  for auto-correlation)
- Images are mostly smooth
  - Basis for compression



Slide credit: J. Hays

## Practice question

1. Match the spatial domain image to the Fourier magnitude image



Slide credit: J. Hays

## Summary

- Frequency domain techniques
- Images in terms of frequency
- Fourier Series
- Convolution Theorem

## Next Week

- Sampling
- Gabor wavelets
- Steerable filters