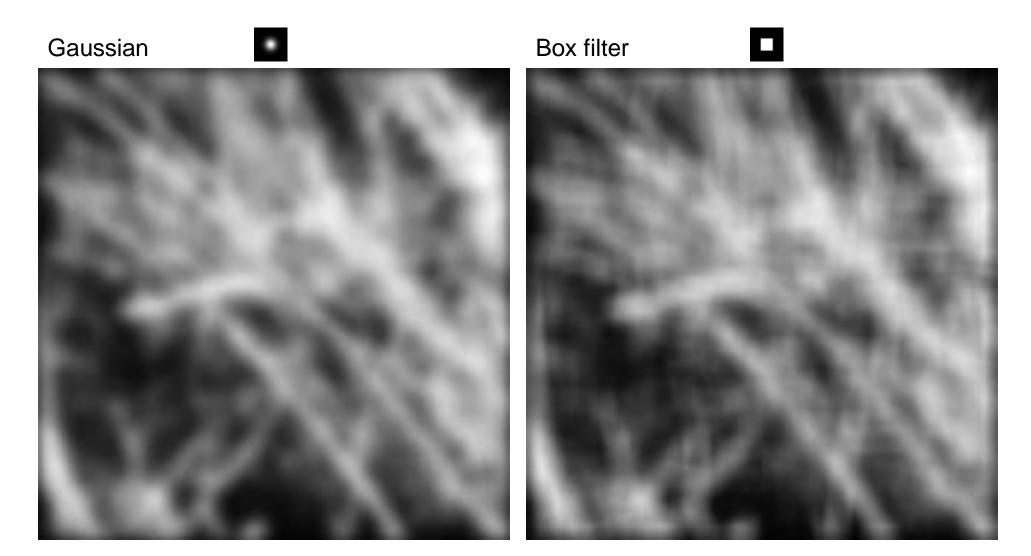
## Fourier Transform and Frequency Domain

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



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



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



#### Answer to these questions?

Thinking images in terms of frequency

Treat images as infinite-size, continuous periodic functions

- Fourier transform and frequency domain
  - Frequency view of filtering
  - Sampling

#### 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.

#### 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

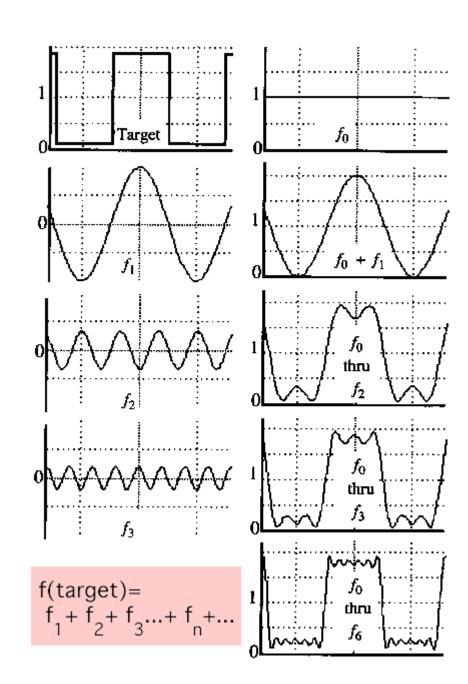


#### A sum of sines

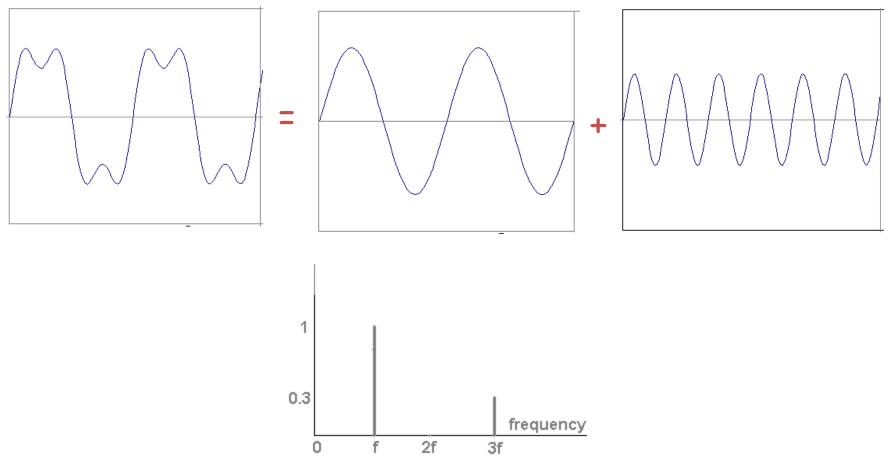
• Our building block:

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

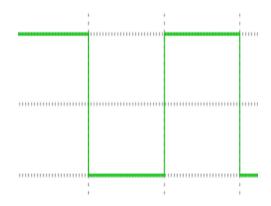
 Add enough of them to get any signal f(x) you want!

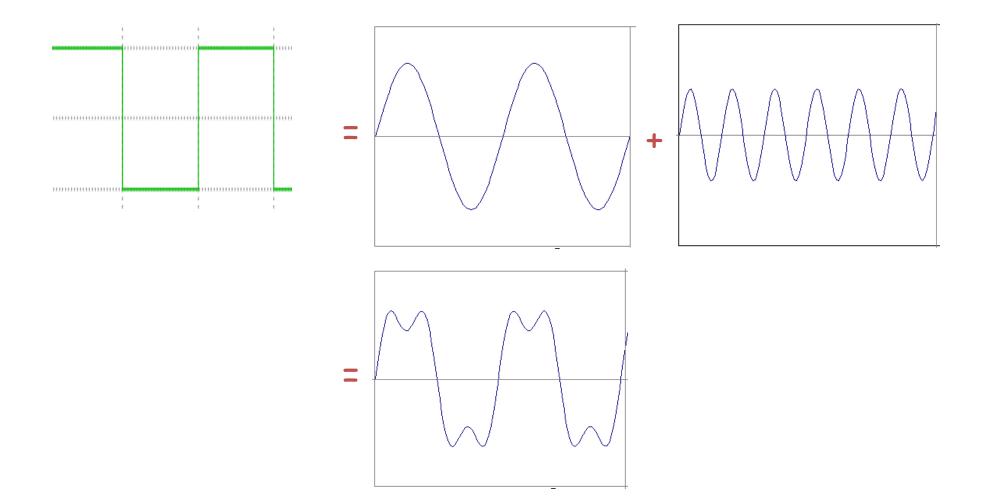


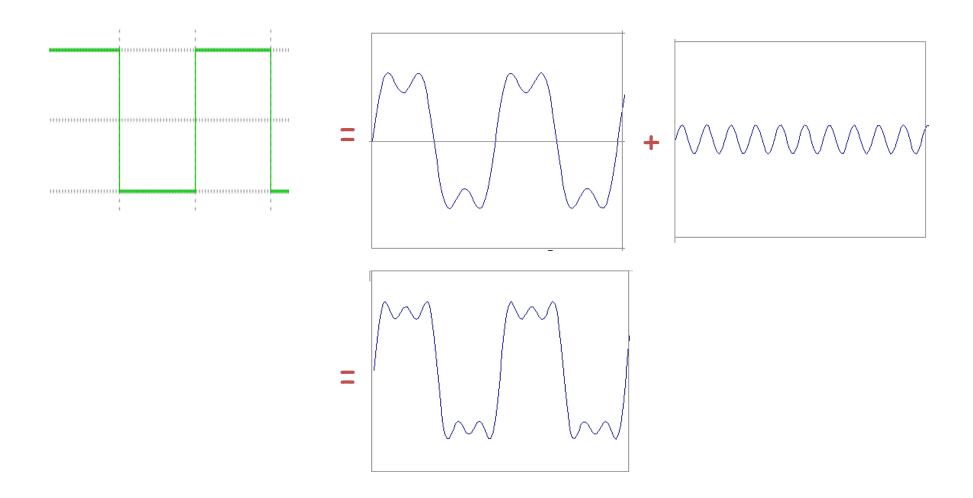
• example :  $g(t) = \sin(2\pi f t) + (1/3)\sin(2\pi(3f) t)$ 

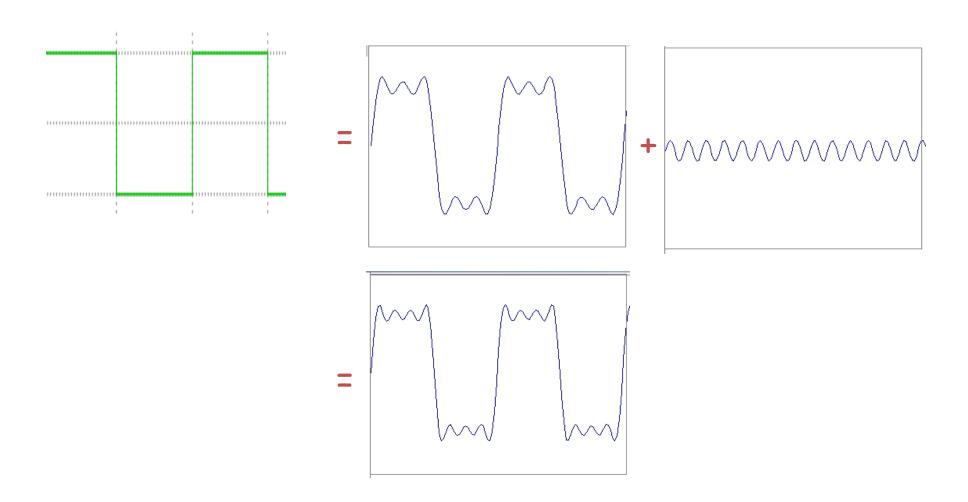


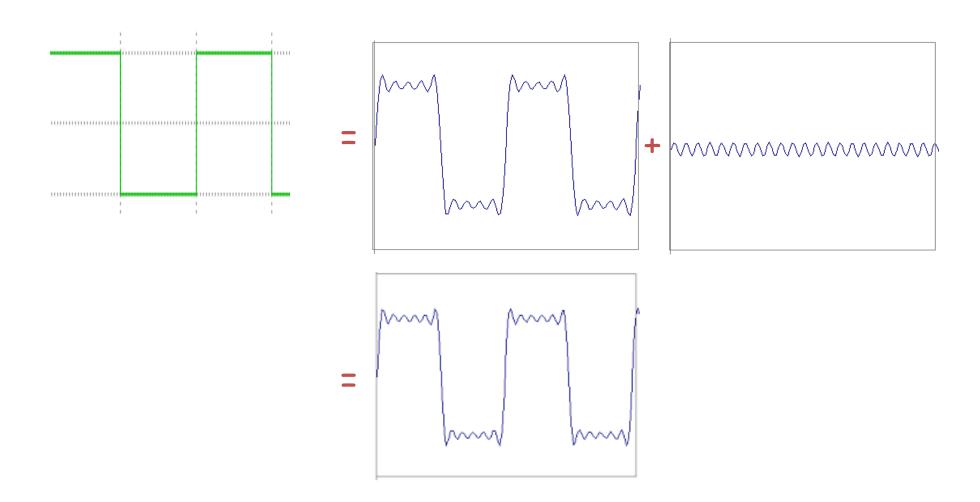
Slides: Efros

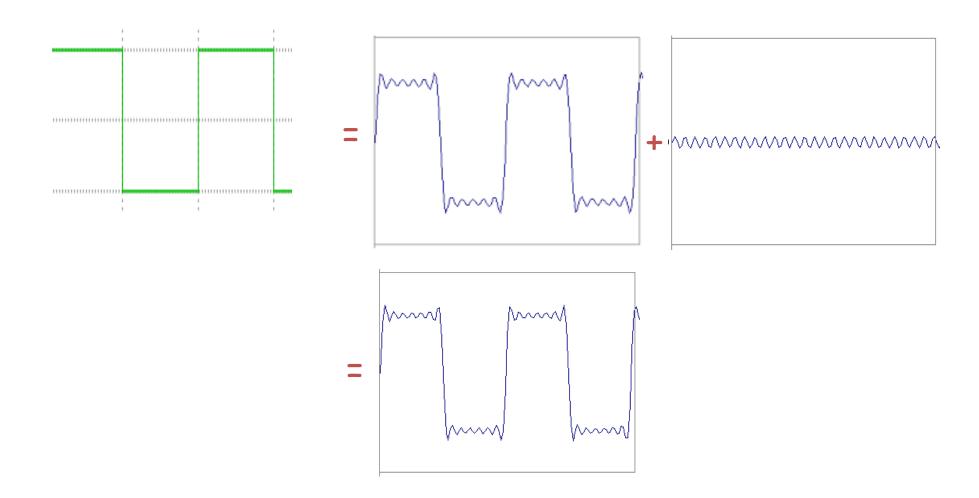


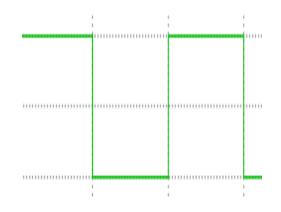




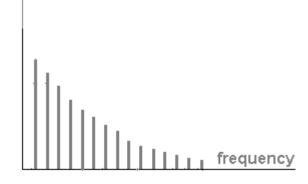






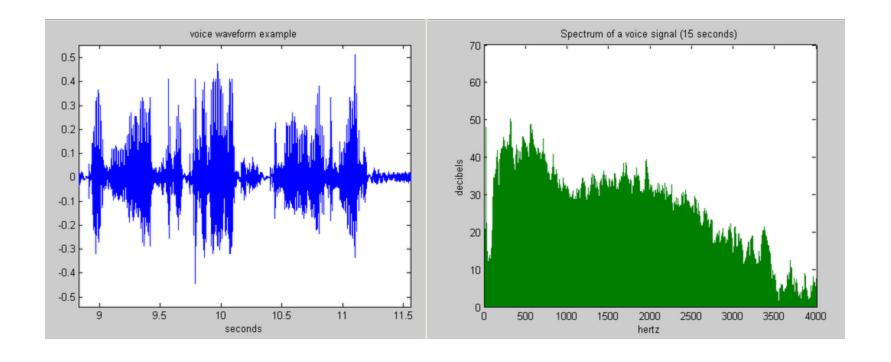


$$A\sum_{k=1}^{\infty} \frac{1}{k} \sin(2\pi kt)$$



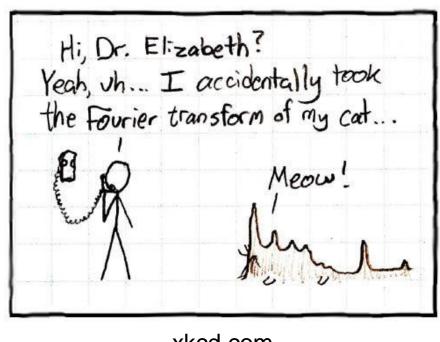
## Example: Music

 We think of music in terms of frequencies at different magnitudes



#### Other signals

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



xkcd.com

#### **Spectral Techniques**

- Technique for representation and analysis of signals in frequency domain including audio, images, video
- How to decompose signal into summation of a series of sine and cosine functions (also called harmonic functions)
- Spectral techniques can improve efficiency of image processing
- Some image processing effects, concepts, techniques easier in frequency domain
- Includes fourier transform, discrete fourier transform and discrete cosine transform

#### **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

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

Euler's formula:  $e^{inx} = \cos(nx) + i\sin(nx)$ 

#### Computing the Fourier Transform

$$H(\omega) = \mathcal{F}\{h(x)\} = Ae^{j\phi}$$

#### Continuous

$$H(\omega) = \int_{-\infty}^{\infty} h(x)e^{-j\omega x}dx$$

#### Discrete

$$H(k) = \frac{1}{N} \sum_{x=0}^{N-1} h(x)e^{-j\frac{2\pi kx}{N}} \qquad k = -N/2..N/2$$

Fast Fourier Transform (FFT): NlogN

#### Fourier analysis in images

- You can consider an image as a signal which is sampled in two directions
- Taking Fourier transform in both X and Y directions gives you the frequency representation of image
  - More intuitively, for the sinusoidal signal, if the amplitude varies so fast in short time, you can say it is a high frequency signal
  - If it varies slowly, it is a low frequency signal
- You can extend the same idea to images
  - Where does the amplitude varies drastically in images?
    - At the edge points, or noises.
    - So we can say, edges and noises are high frequency contents in an image
  - If there is no much changes in amplitude, it is a low frequency component

#### Discrete Fourier Transform (DFT) for image (1)

- The DFT is the sampled Fourier Transform and therefore does not contain all frequencies forming an image, but only a set of samples which is large enough to fully describe the spatial domain image
- The number of frequencies corresponds to the number of pixels in the spatial domain image, *i.e.* the image in the spatial and Fourier domain are of the same size

#### Discrete Fourier Transform (DFT) for image (2)

For a square image of size N×N, the two-dimensional DFT is given by:

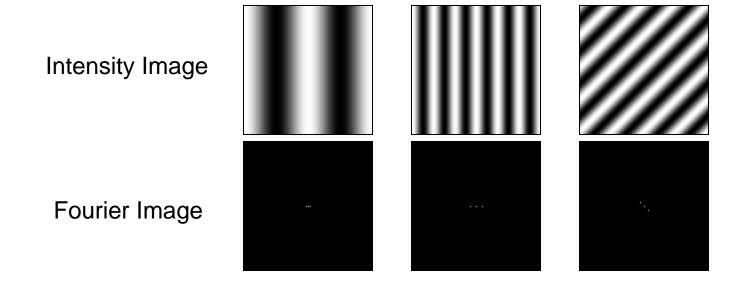
$$F(k,l) = \sum_{i=0}^{N-1} \sum_{j=0}^{N-1} f(i,j) e^{-\iota 2\pi (\frac{ki}{N} + \frac{lj}{N})}$$

- where f(a,b) is the image in the spatial domain and the exponential term is the basis function corresponding to each point F(k,l) in the Fourier space
- The equation can be interpreted as: the value of each point F(k,l) is obtained by multiplying the spatial image with the corresponding base function and summing the result
- The basis functions are sine and cosine waves with increasing frequencies, i.e. F(0,0) represents the DC-component of the image which corresponds to the average brightness and F(N-1,N-1) represents the highest frequency

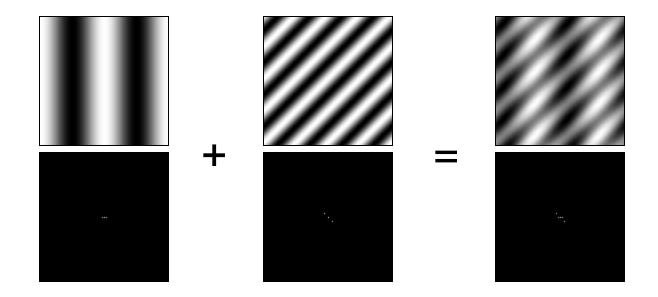
#### Discrete Fourier Transform (DFT) for image (3)

- The Fourier Transform is used if we want to access the geometric characteristics of a spatial domain image
  - Because the image in the Fourier domain is decomposed into its sinusoidal components, it is easy to examine or process certain frequencies of the image, thus influencing the geometric structure in the spatial domain
- In most implementations the Fourier image is shifted in such a way that the DC-value (i.e. the image mean) F(0,0) is displayed in the center of the image. The further away from the center an image point is, the higher is its corresponding frequency
- The DC-value is by far the largest component of the image. However, the
  dynamic range of the Fourier coefficients (i.e. the intensity values in the Fourier
  image) is too large to be displayed on the screen.
- We apply a logarithmic transformation to the image we obtain

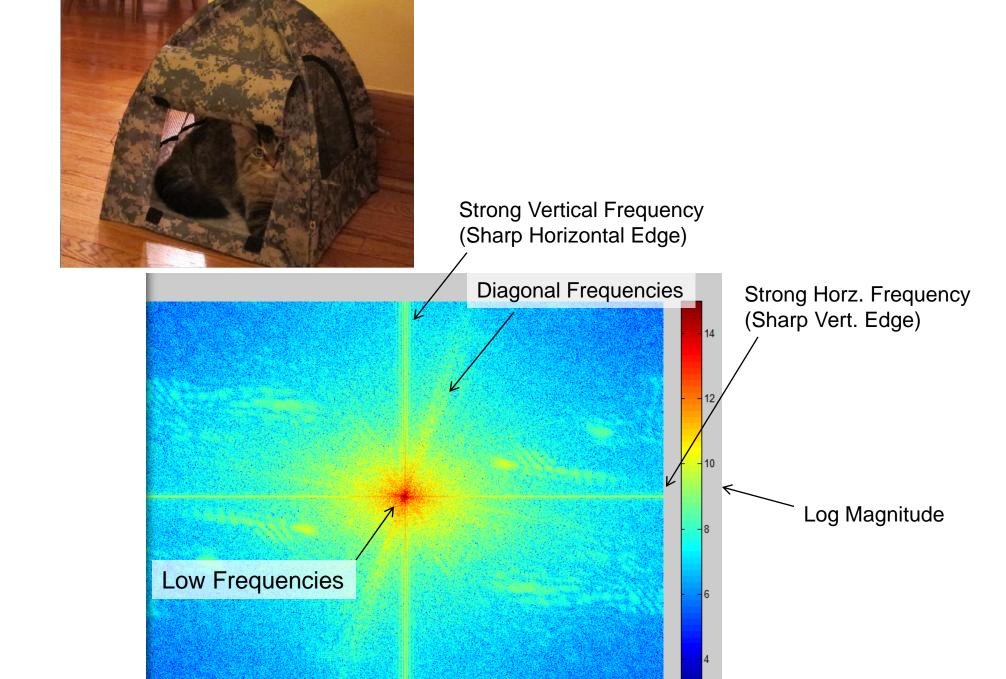
#### Fourier analysis in images



#### Signals can be composed



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



#### 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!

#### **Properties of Fourier Transforms**

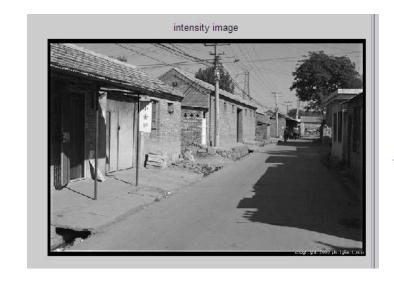
• Linearity 
$$\mathcal{F}[ax(t) + by(t)] = a\mathcal{F}[x(t)] + b\mathcal{F}[y(t)]$$

Fourier transform of a real signal is symmetric about the origin

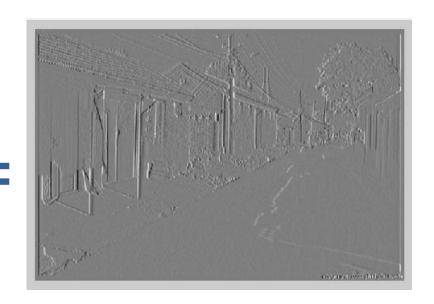
 The energy of the signal is the same as the energy of its Fourier transform

## Filtering in spatial domain

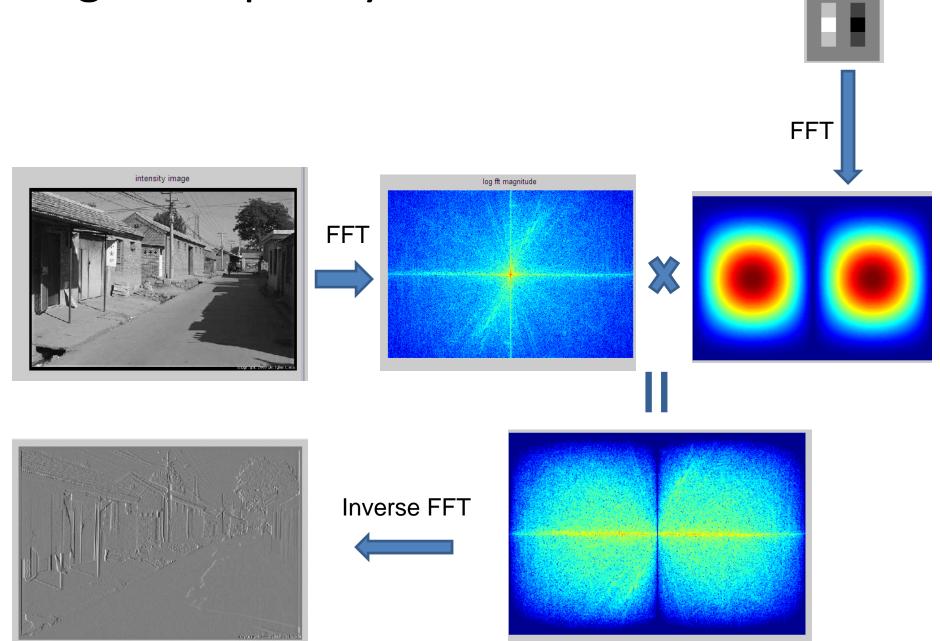
1	0	-1
2	0	-2
1	0	-1





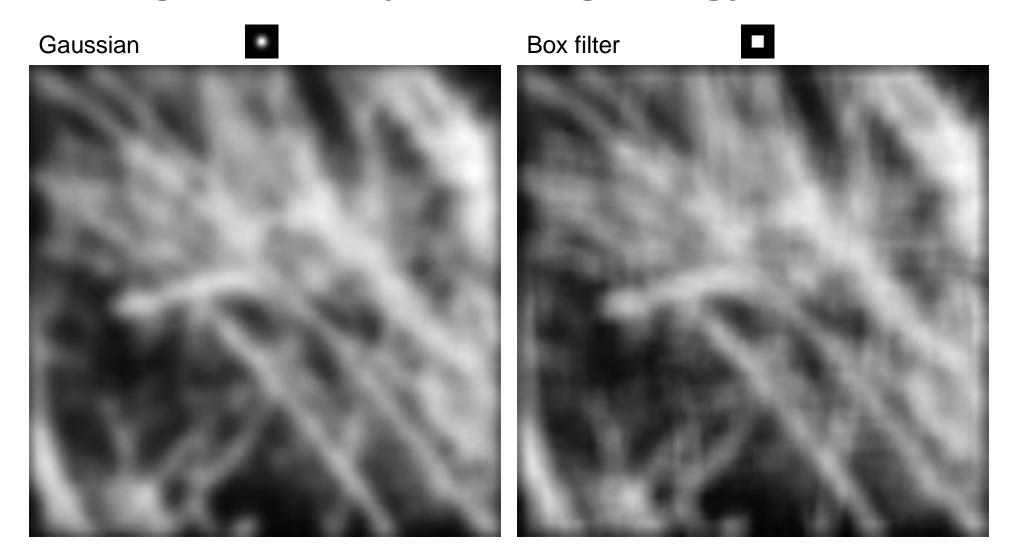


## Filtering in frequency domain

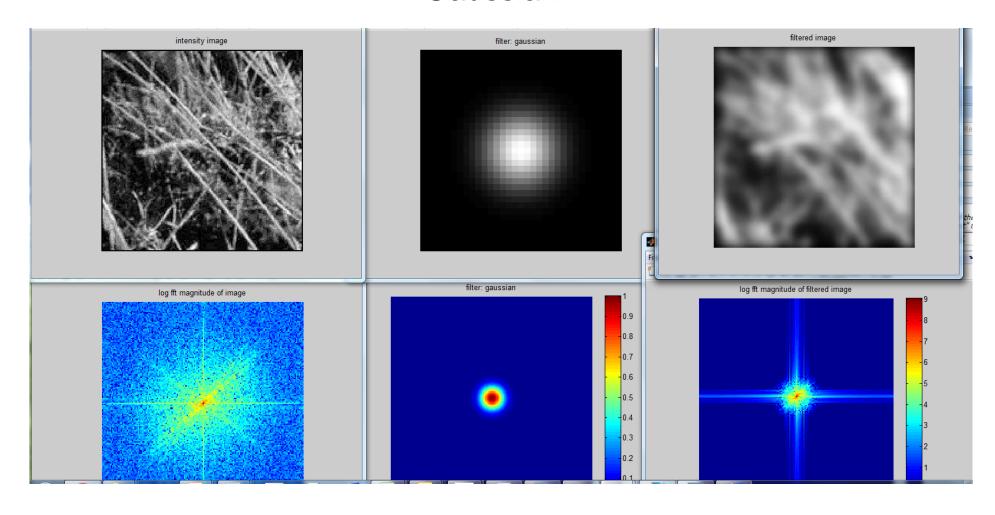


Filtering

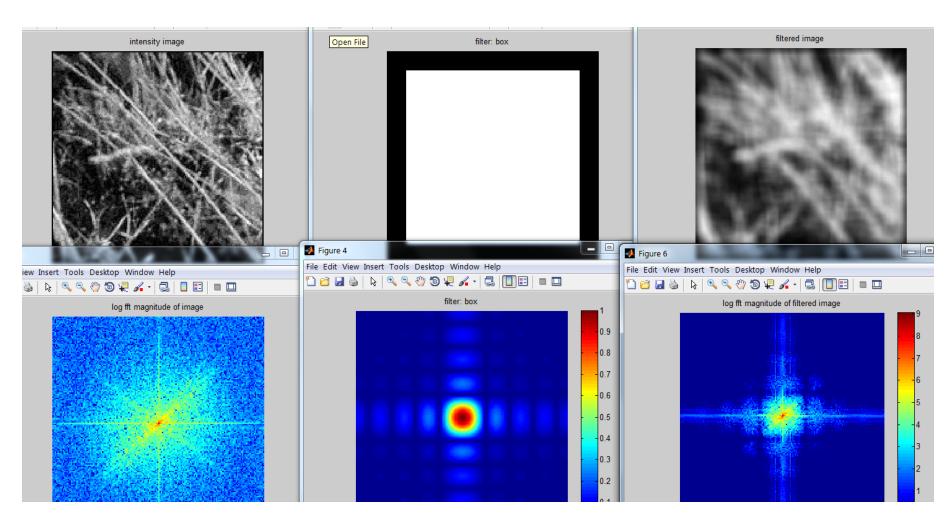
# Why does the Gaussian give a nice smooth image, but the square filter give edgy artifacts?



#### Gaussian

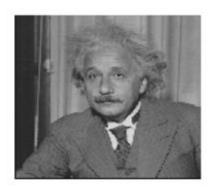


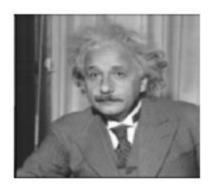
#### **Box Filter**

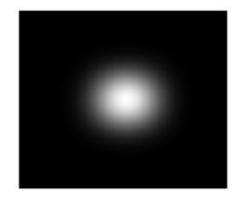


## Low-pass, Band-pass/High-pass filters

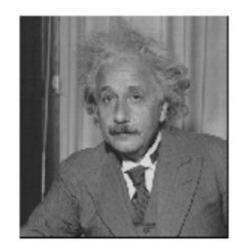
#### low-pass:



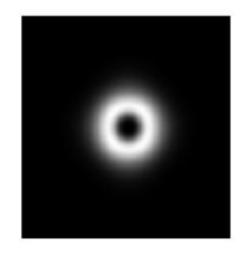




High-pass / band-pass:







## Edges in images



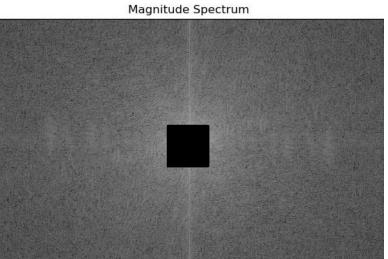


Image after HPF



#### Phase and magnitude

#### Curious fact

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

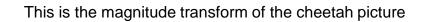
#### Demonstration

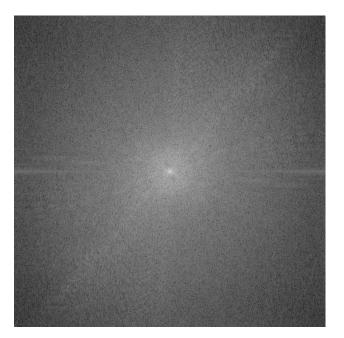
— Take two pictures, swap the phase transforms, compute the inverse — what does the result look like?



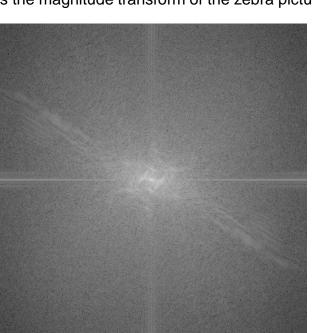




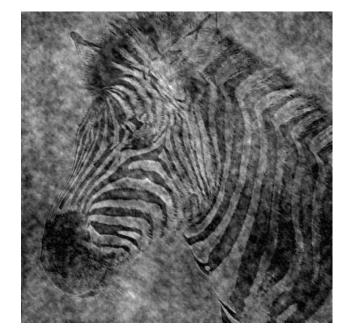




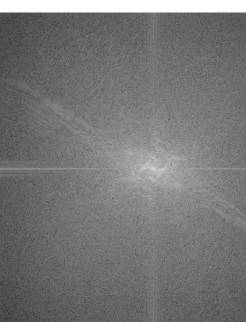
This is the magnitude transform of the zebra picture



Reconstruction with zebra phase, cheetah magnitude



Reconstruction with cheetah phase, zebra magnitude





#### Sampling

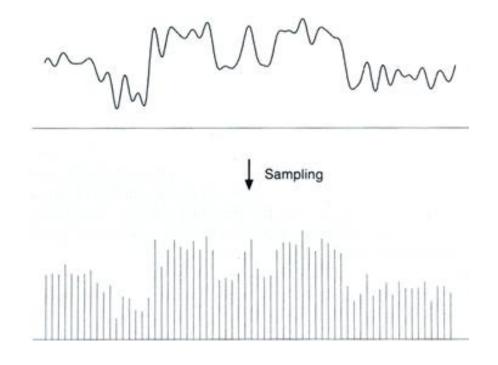
# Why does a lower resolution image still make sense to us? What do we lose?

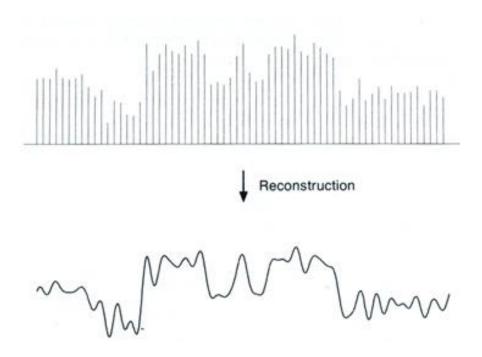


Image: <a href="http://www.flickr.com/photos/igorms/136916757/">http://www.flickr.com/photos/igorms/136916757/</a>

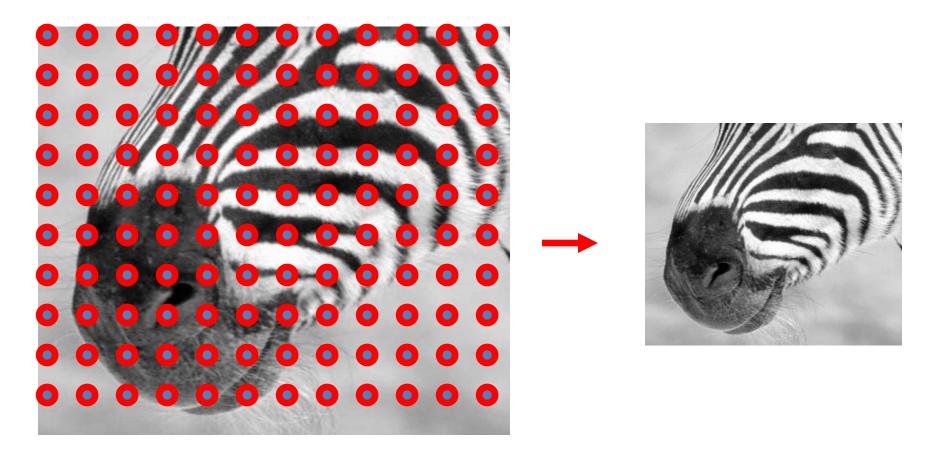
#### Sampled representations

- How to store and compute with continuous functions?
- Common scheme for representation: samples
  - write down the function's values at many points
- Reconstruction: making samples back into a continuous function





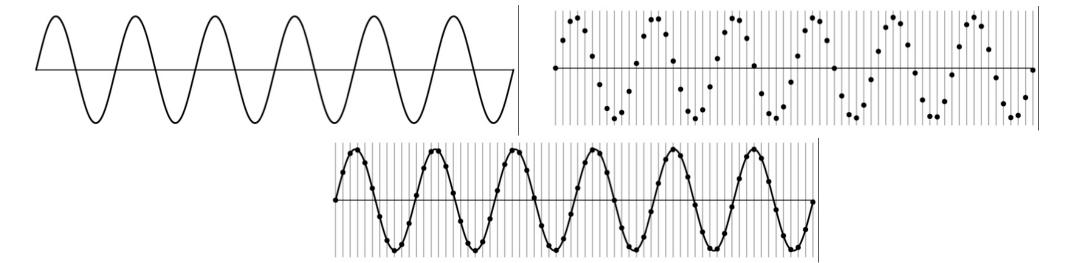
# Subsampling by a factor of 2



Throw away every other row and column to create a 1/2 size image

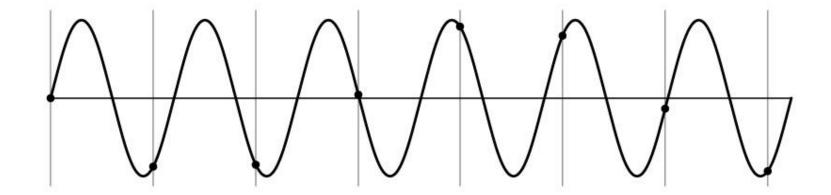
#### Undersampling

- What if we missed things between the samples?
- Simple example: undersampling a sine wave
  - unsurprising result: information is lost
  - surprising result: indistinguishable from lower frequency
  - also was always indistinguishable from higher frequencies
  - aliasing: signals 'traveling in disguise' as other frequencies



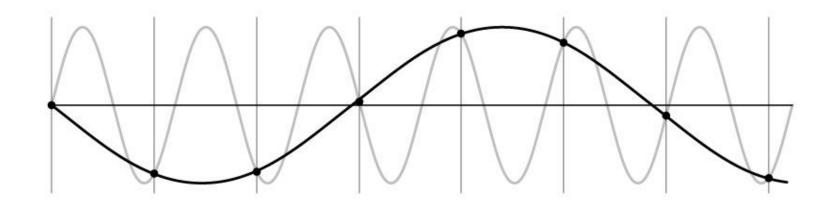
# Aliasing problem

• 1D example (sinewave):



# Aliasing problem

• 1D example (sinewave):



#### Aliasing problem

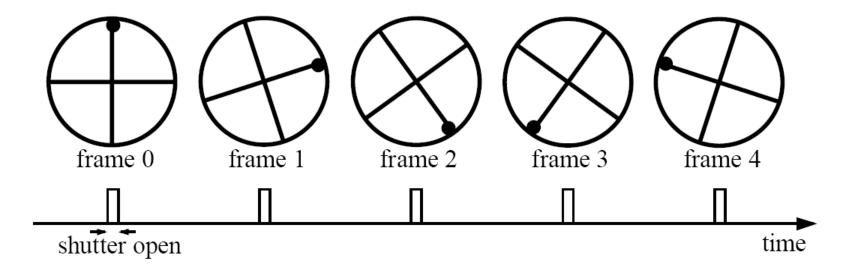
- Sub-sampling may be dangerous....
- Characteristic errors may appear:
  - "Wagon wheels rolling the wrong way in movies"
  - "Checkerboards disintegrate in ray tracing"
  - "Striped shirts look funny on color television"

Source: D. Forsyth

#### Aliasing in video

Imagine a spoked wheel moving to the right (rotating clockwise). Mark wheel with dot so we can see what's happening.

If camera shutter is only open for a fraction of a frame time (frame time = 1/30 sec. for video, 1/24 sec. for film):

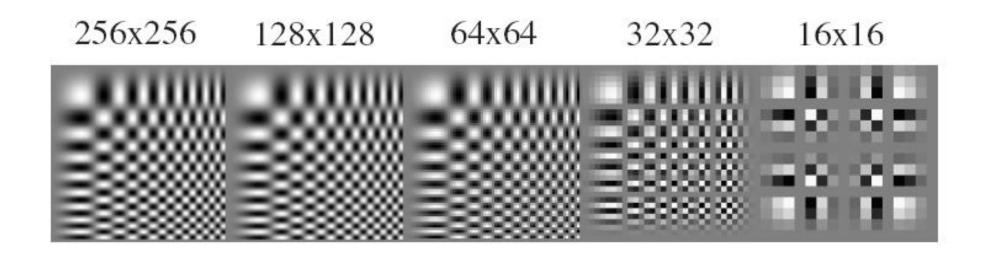


Without dot, wheel appears to be rotating slowly backwards! (counterclockwise)

## Aliasing in graphics

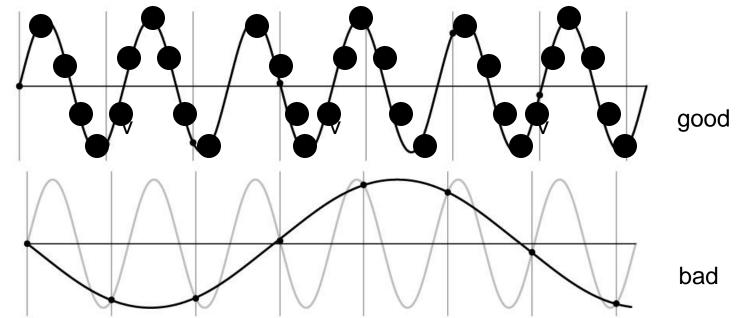


# Sampling and aliasing



#### Nyquist-Shannon Sampling Theorem

- When sampling a signal at discrete intervals, the sampling frequency must be  $\geq 2 \times f_{max}$
- f<sub>max</sub> = max frequency of the input signal
- This will allows to reconstruct the original perfectly from the sampled version



### Anti-aliasing

#### **Solutions:**

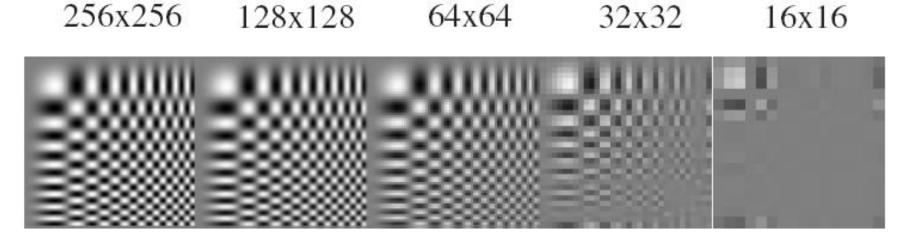
Sample more often

- Get rid of all frequencies that are greater than half the new sampling frequency
  - Will lose information
  - But it's better than aliasing
  - Apply a smoothing filter

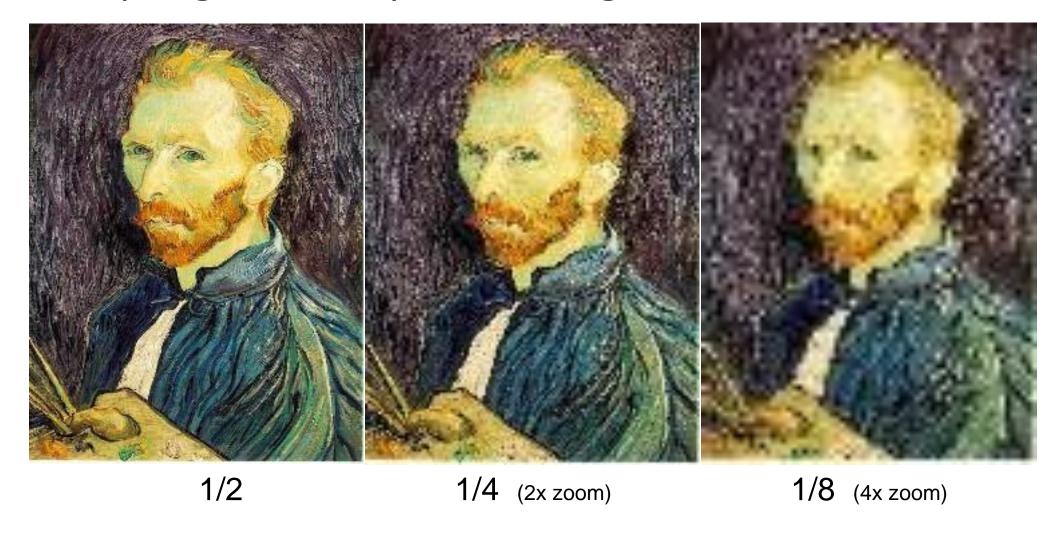
# Anti-aliasing

256x256 128x128 64x64 32x32 16x16

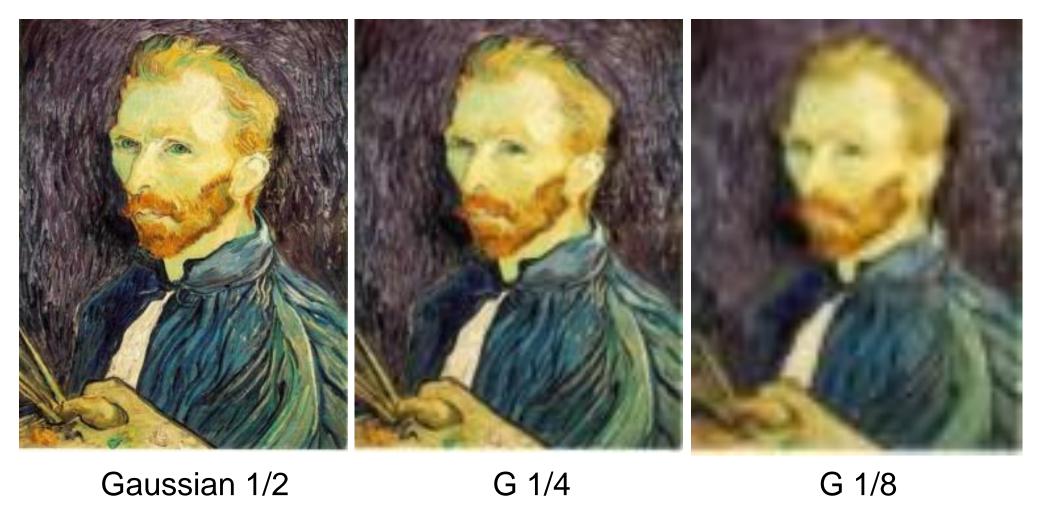
Apply low-pass filter and Sample every other pixel



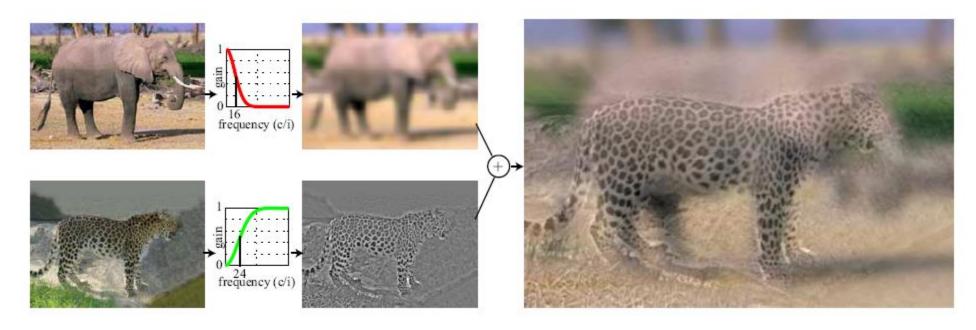
### Subsampling without pre-filtering



## Subsampling with Gaussian pre-filtering



### **Hybrid Images**

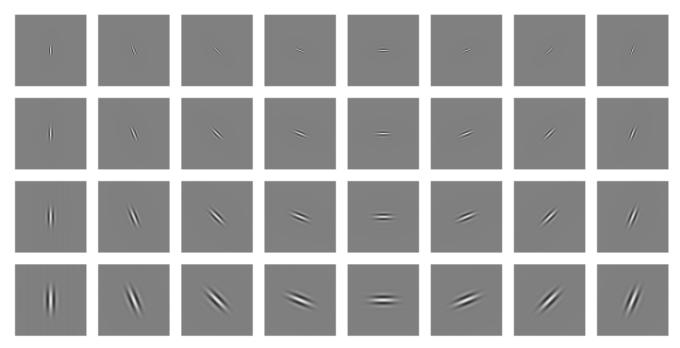




 A. Oliva, A. Torralba, P.G. Schyns, <u>"Hybrid Images,"</u> SIGGRAPH 2006

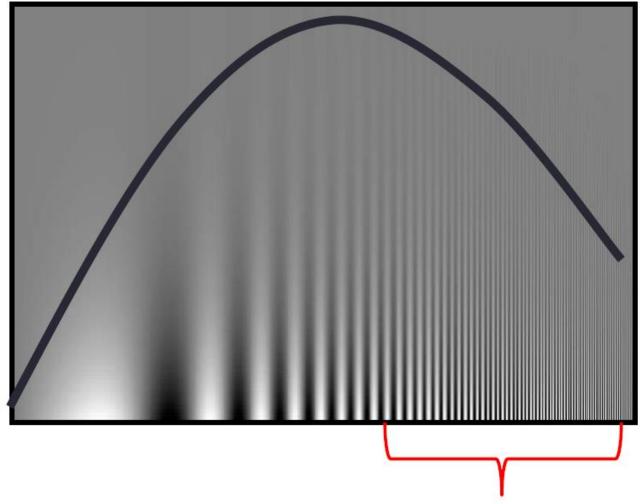
#### Clues from human perception

- 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



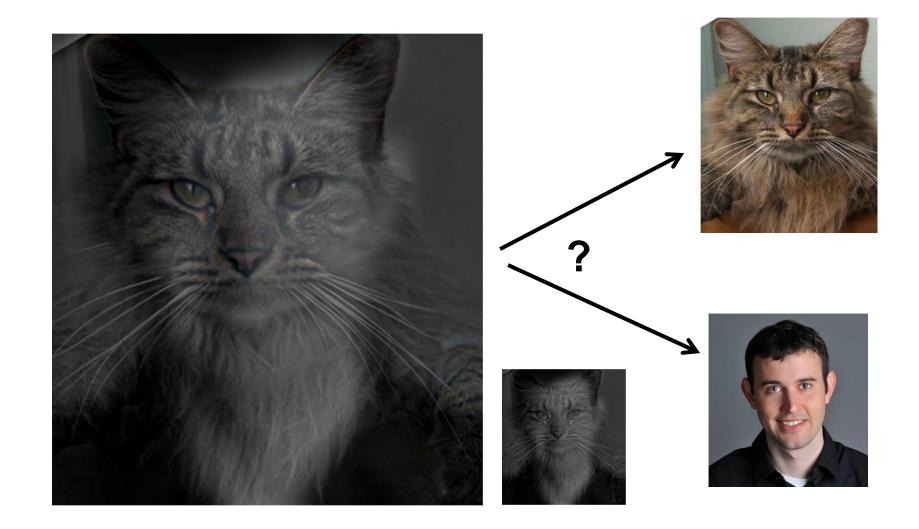
Early Visual Processing: Multi-scale edge and blob filters

# Campbell-Robson contrast sensitivity curve



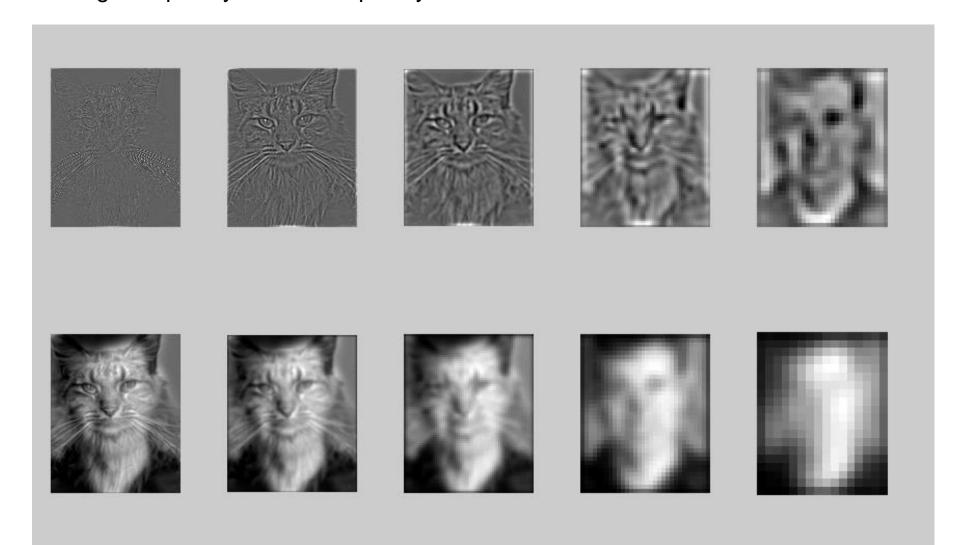
The higher the frequency the less sensitive human visual system is...

# Why do we get different, distance-dependent interpretations of hybrid images?

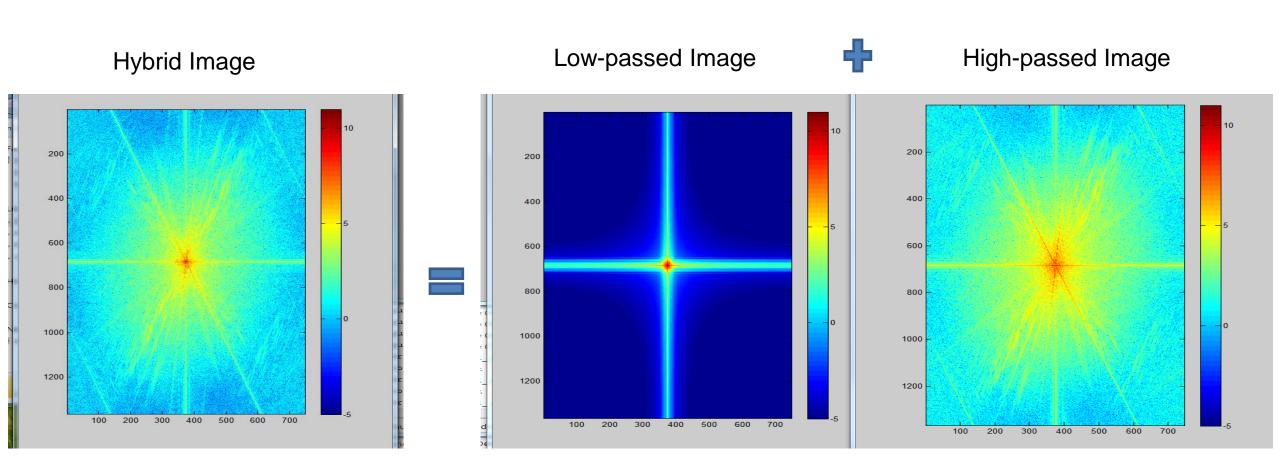


# Hybrid Image in Laplacian Pyramid

High frequency → Low frequency

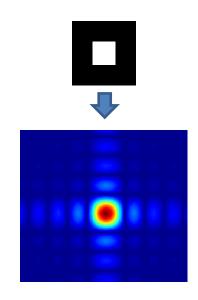


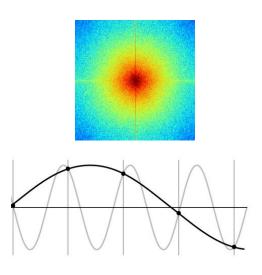
# Hybrid Image in FFT



#### 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 logN vs. N<sup>2</sup> for convolution)
- Images are mostly smooth
  - Basis for compression
- Remember to low-pass before sampling





#### Practice question

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

