

# Filters (cont.)

CS 554 – Computer Vision

Pinar Duygulu

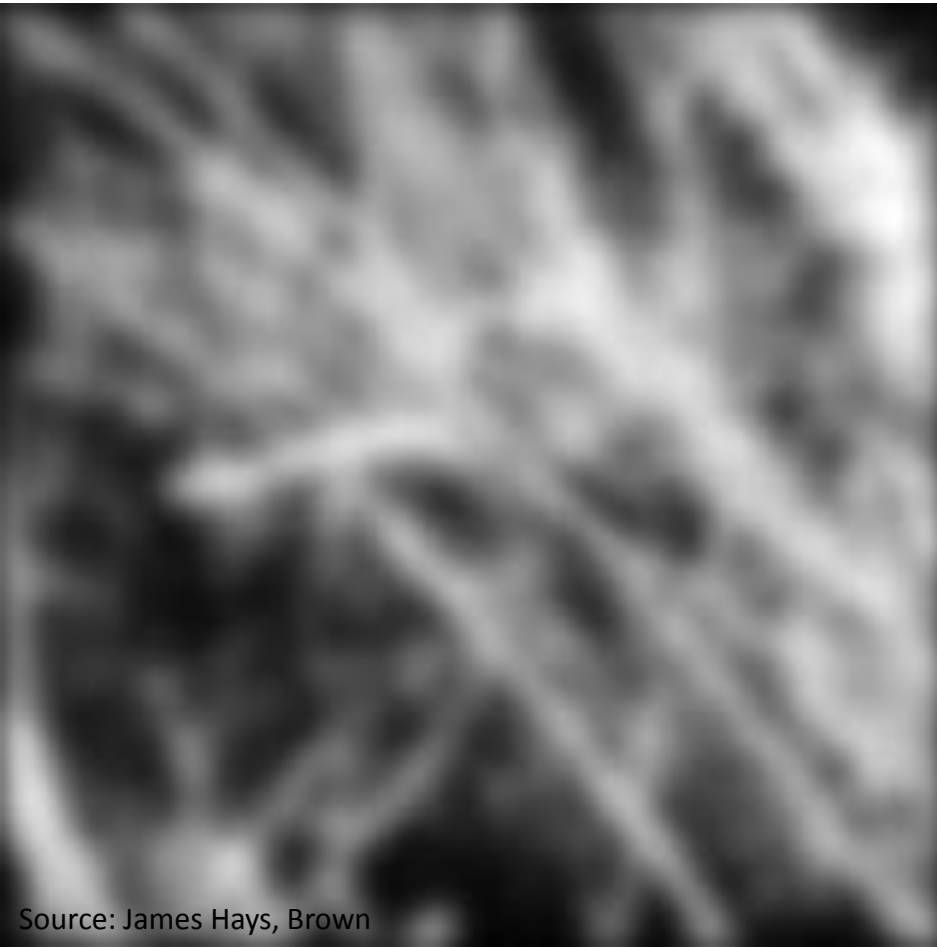
Bilkent University

# Today's topics

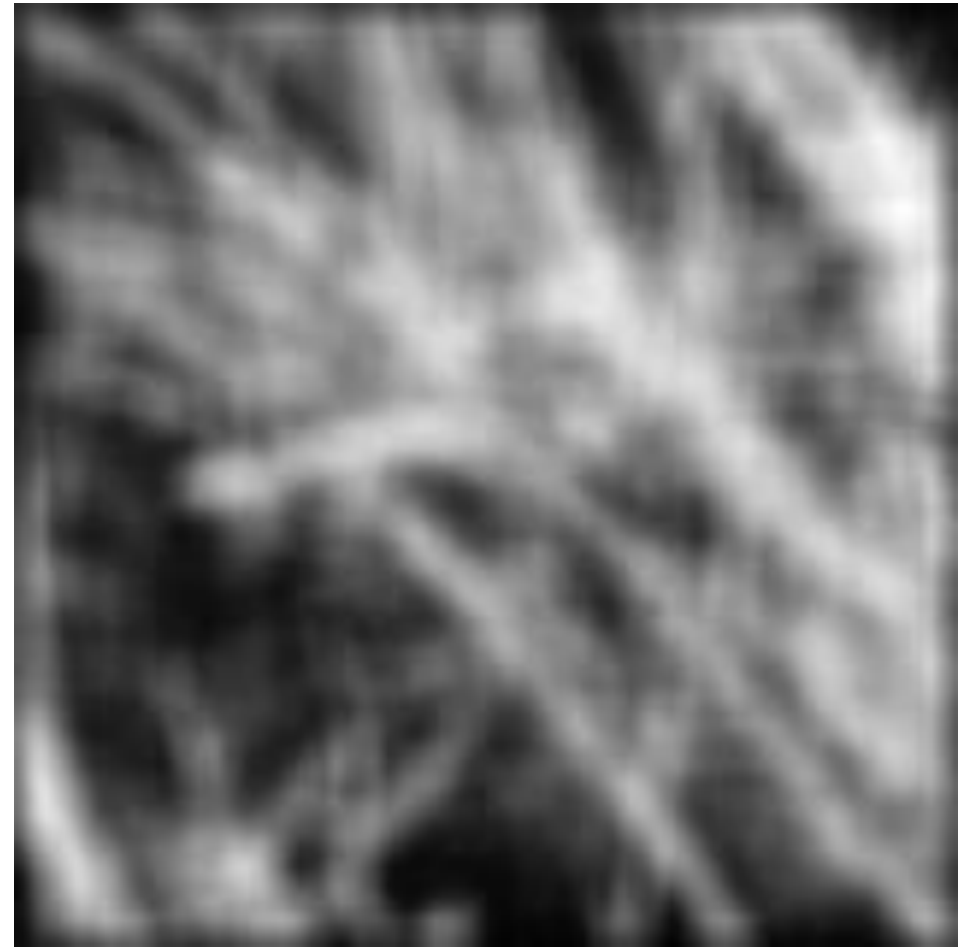
- Image Formation
- Image filters in spatial domain
  - Filter is a mathematical operation of a grid of numbers
  - Smoothing, sharpening, measuring texture
- Image filters in the frequency domain
  - Filtering is a way to modify the frequencies of images
  - Denoising, sampling, image compression
- Templates and Image Pyramids
  - Filtering is a way to match a template to the image
  - Detection, coarse-to-fine registration

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

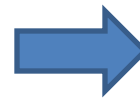
Gaussian



Box filter



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



# Thinking in terms of frequency

# Jean Baptiste Joseph Fourier (1768-

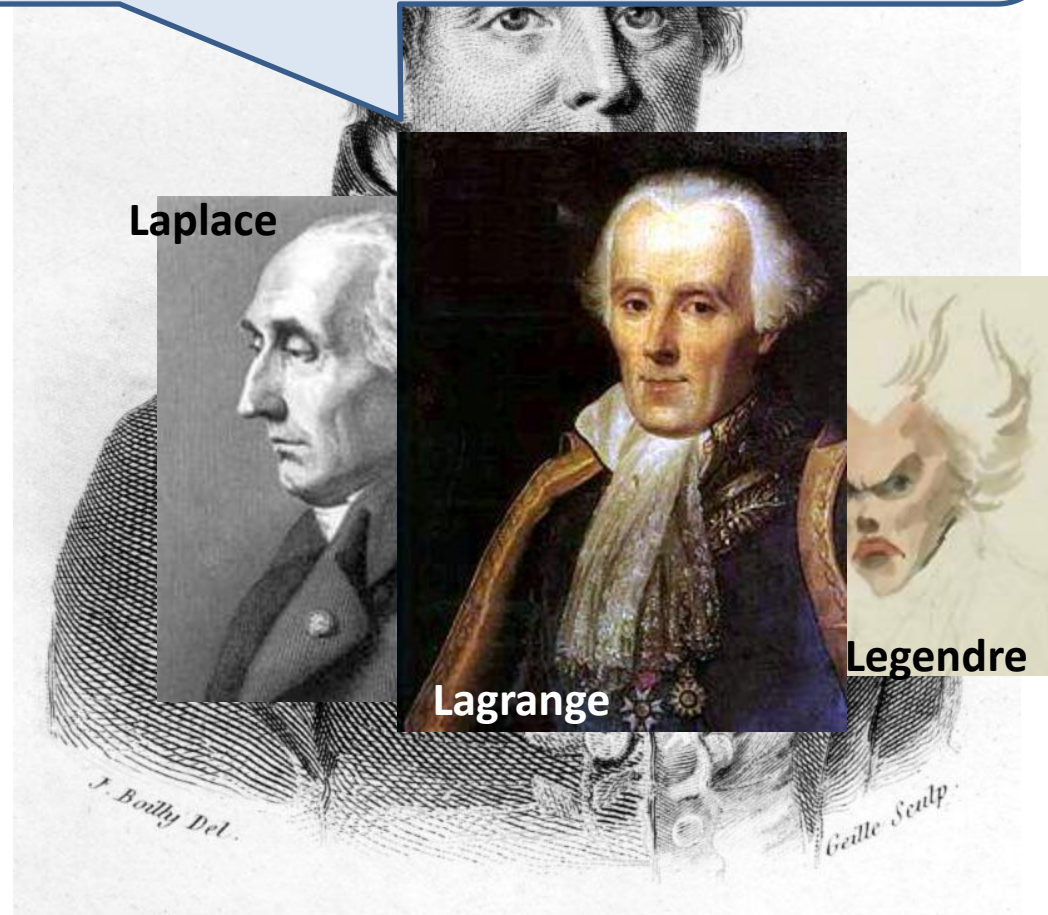
1830)

had crazy idea (1807):

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

*...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!
- But it's (mostly) true!
  - called Fourier Series
  - there are some subtle restrictions

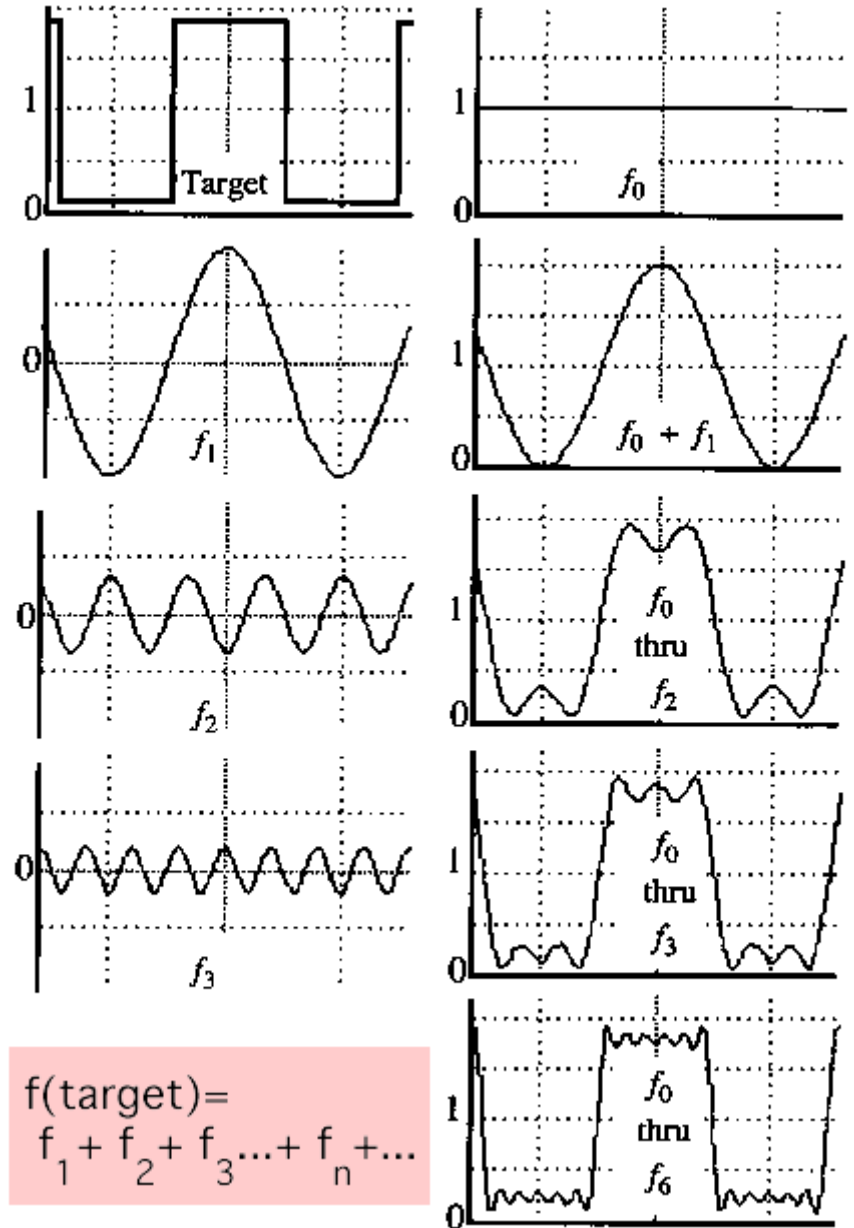


# A sum of sines

Our building block:

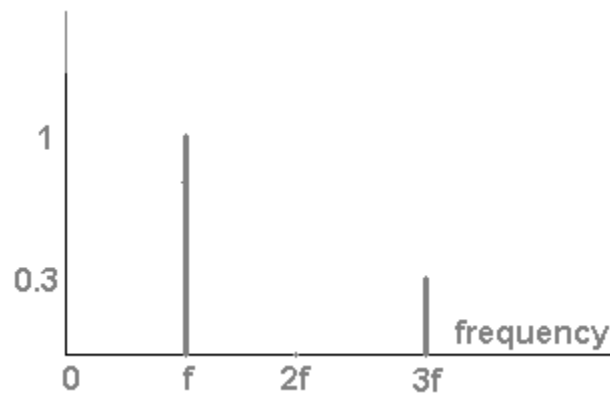
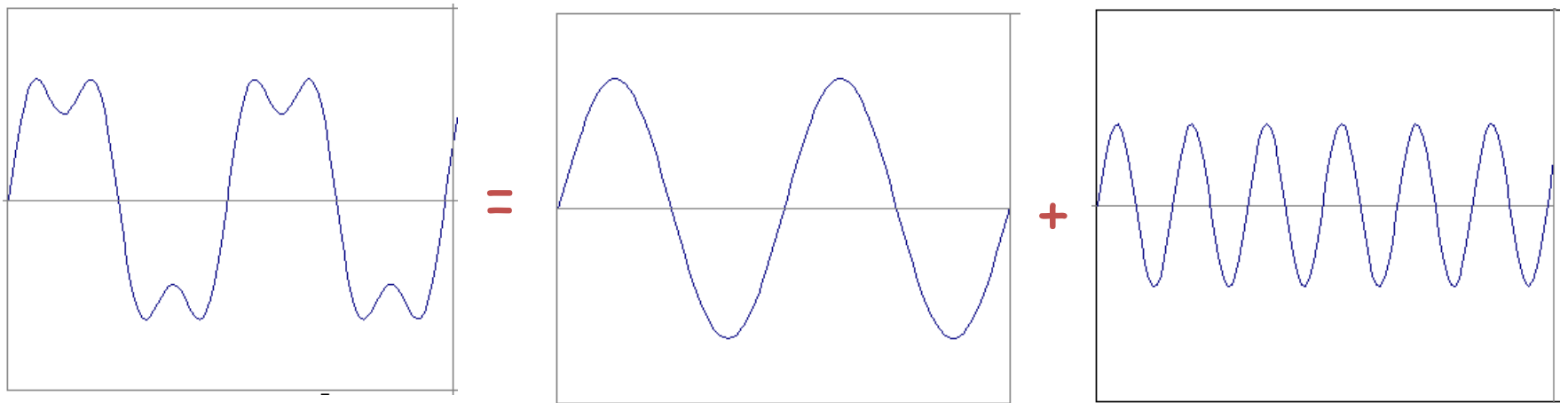
$$A \sin(\omega x + \phi)$$

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



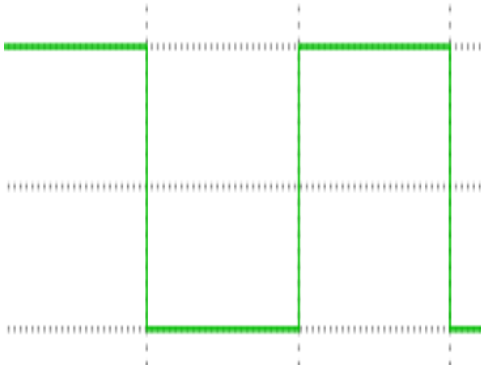
# Frequency Spectra

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

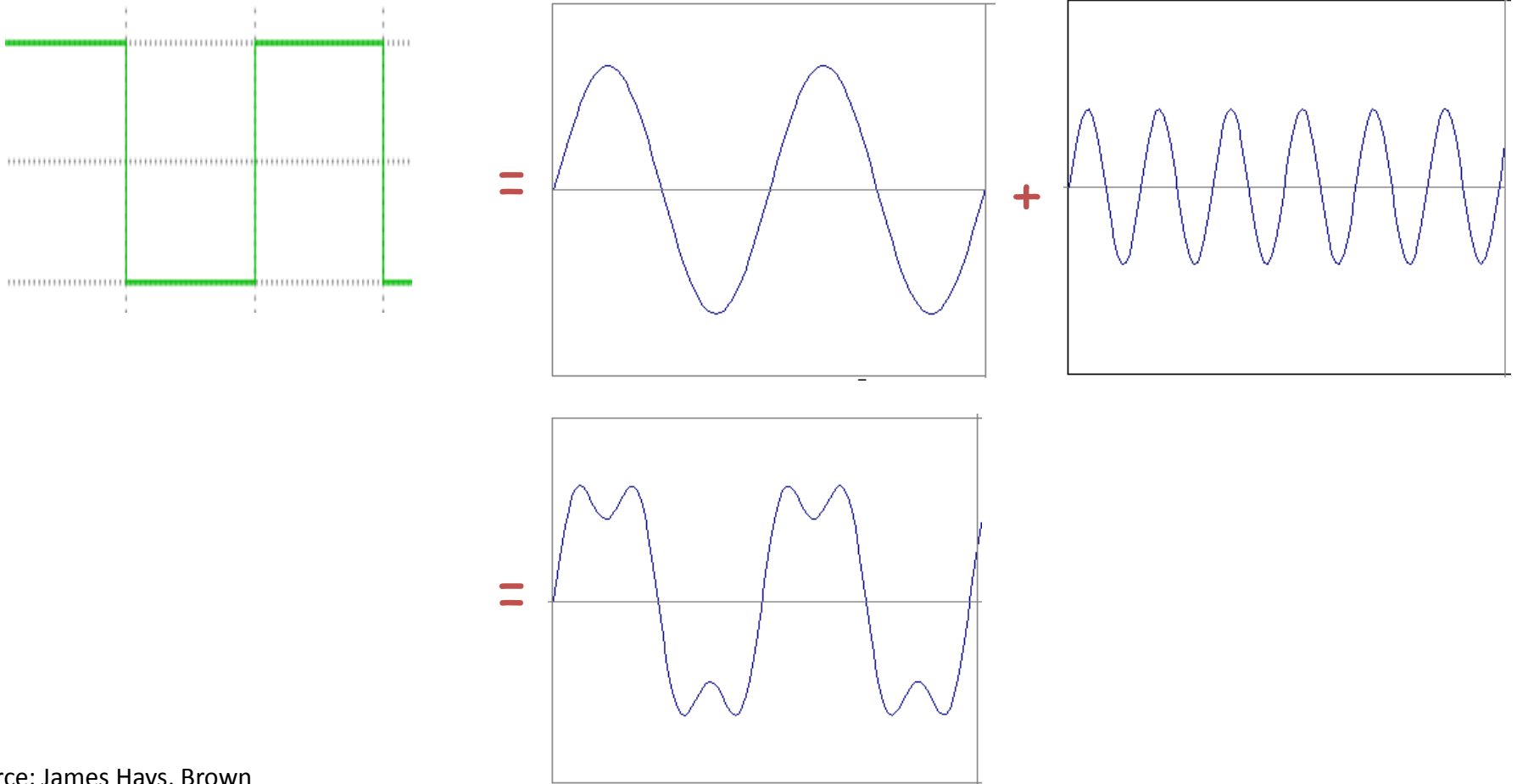




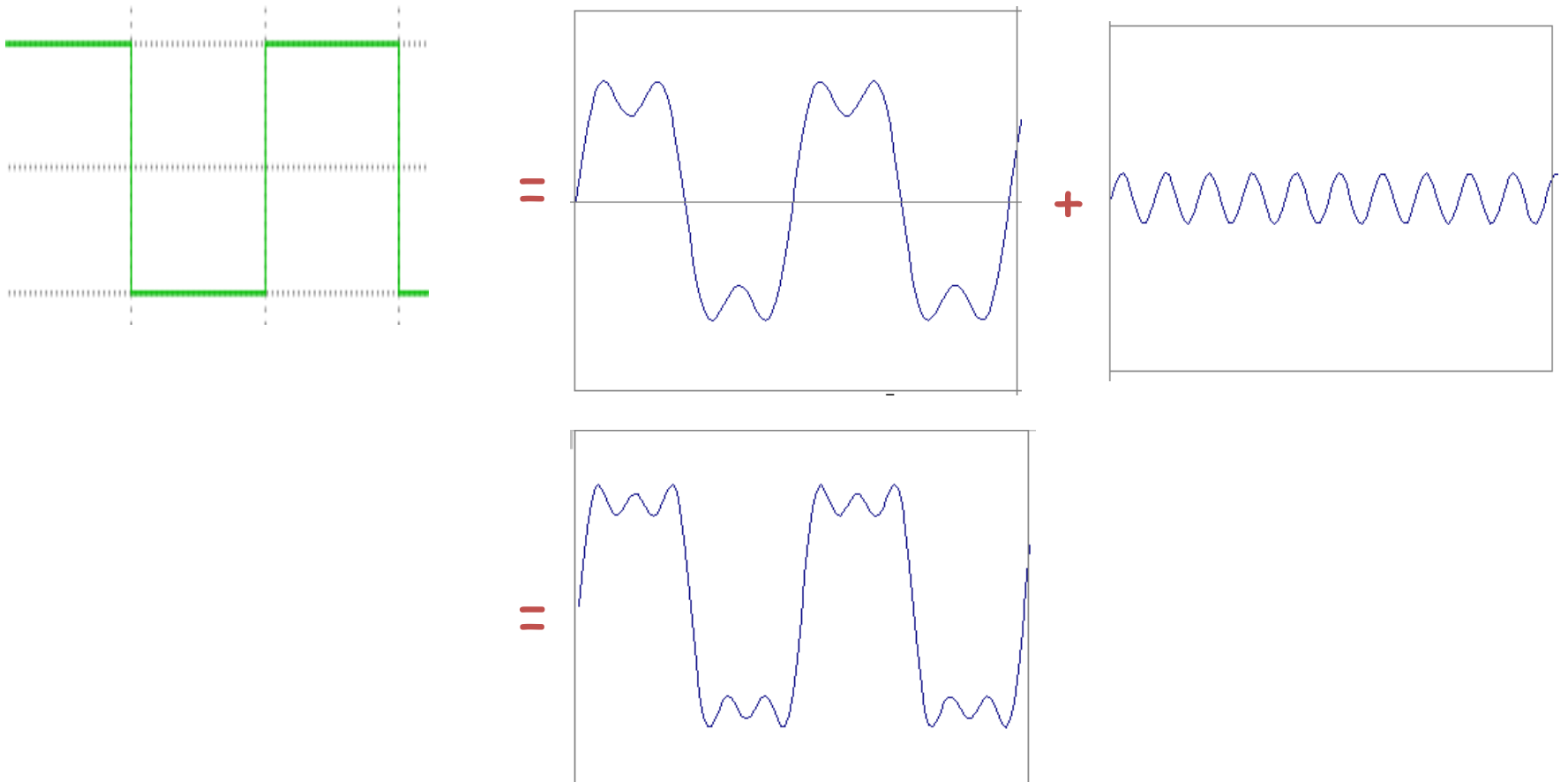
# Frequency Spectra



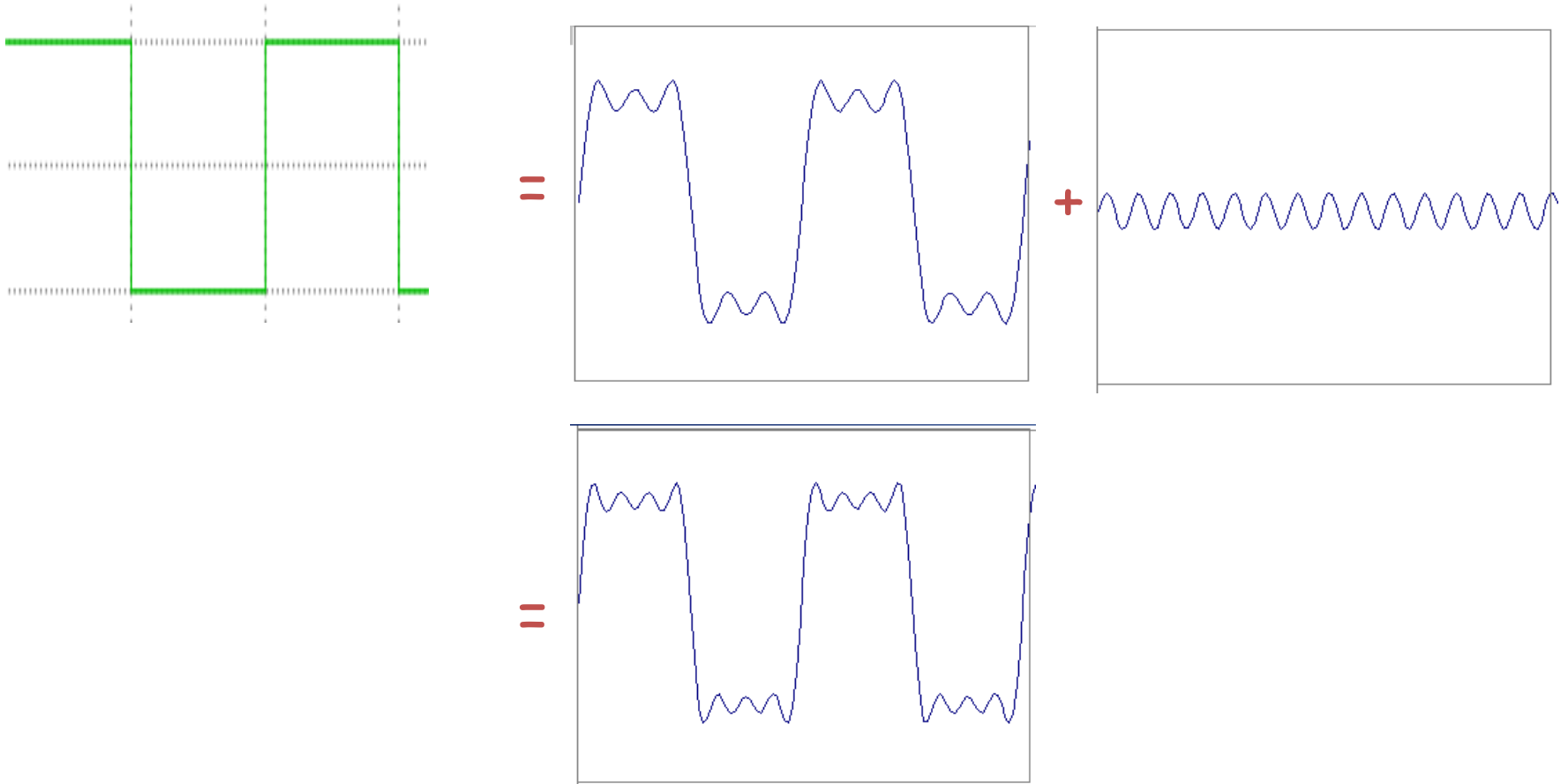
# Frequency Spectra



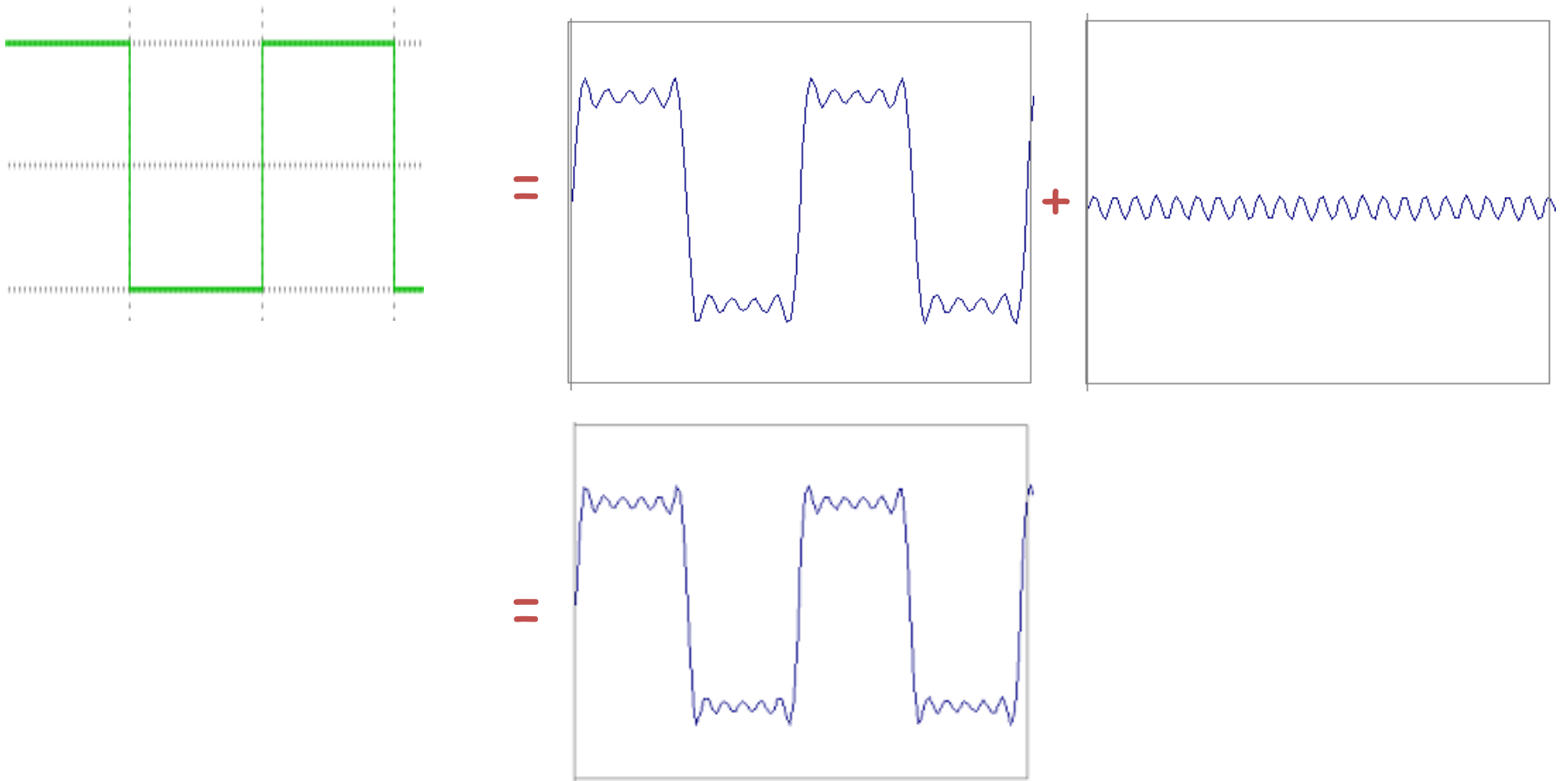
# Frequency Spectra



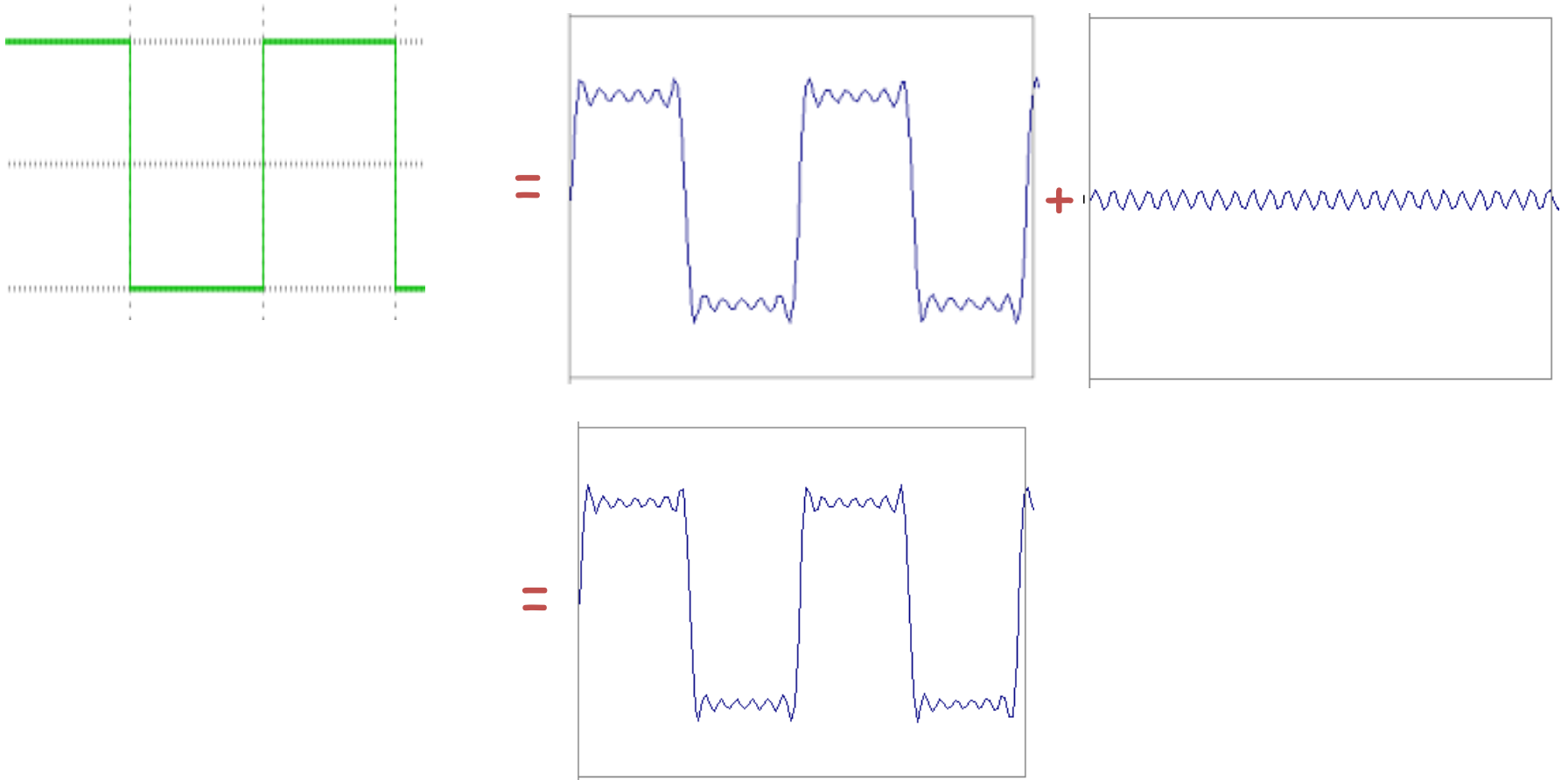
# Frequency Spectra



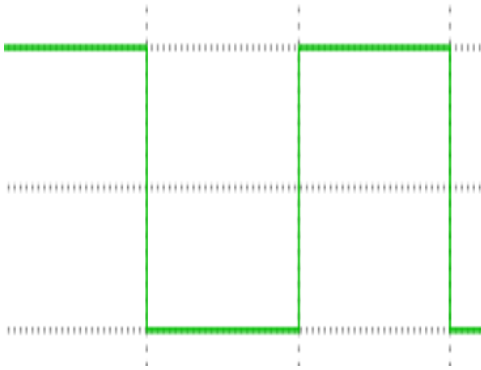
# Frequency Spectra



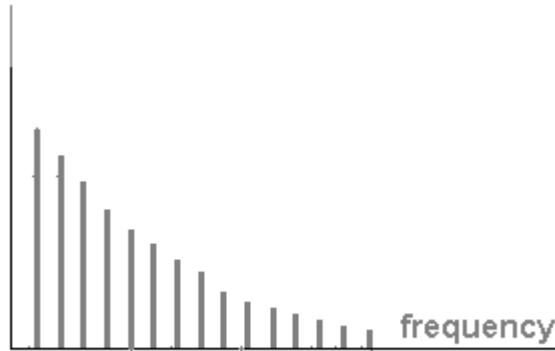
# Frequency Spectra



# Frequency Spectra

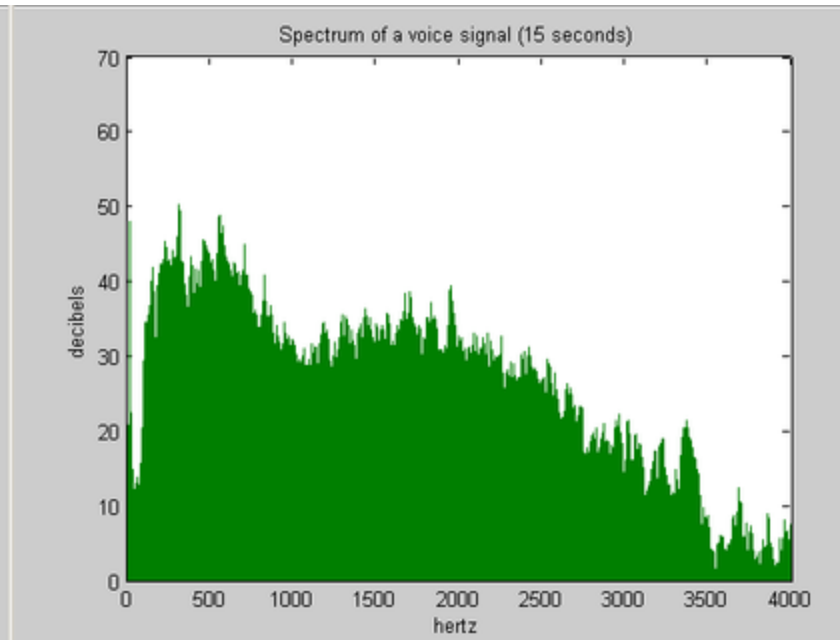
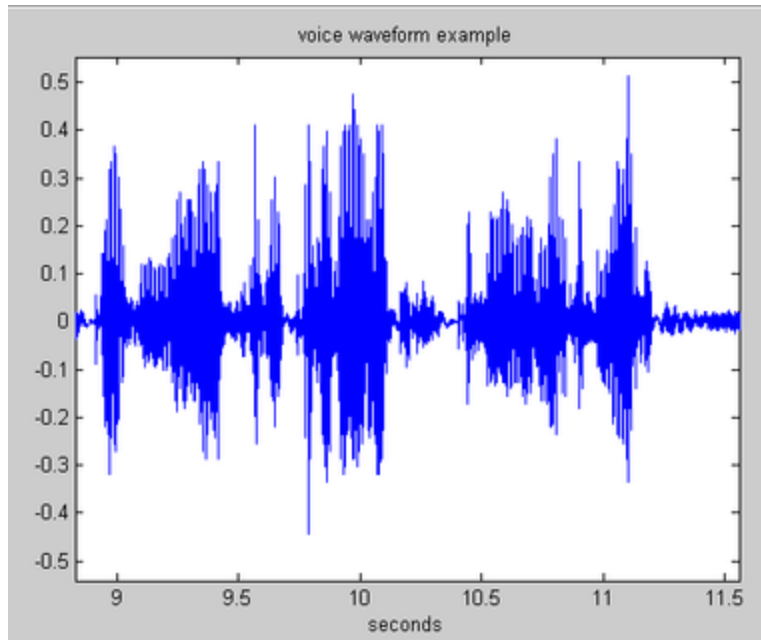


$$= A \sum_{k=1}^{\infty} \frac{1}{k} \sin(2\pi kt)$$



# Example: Music

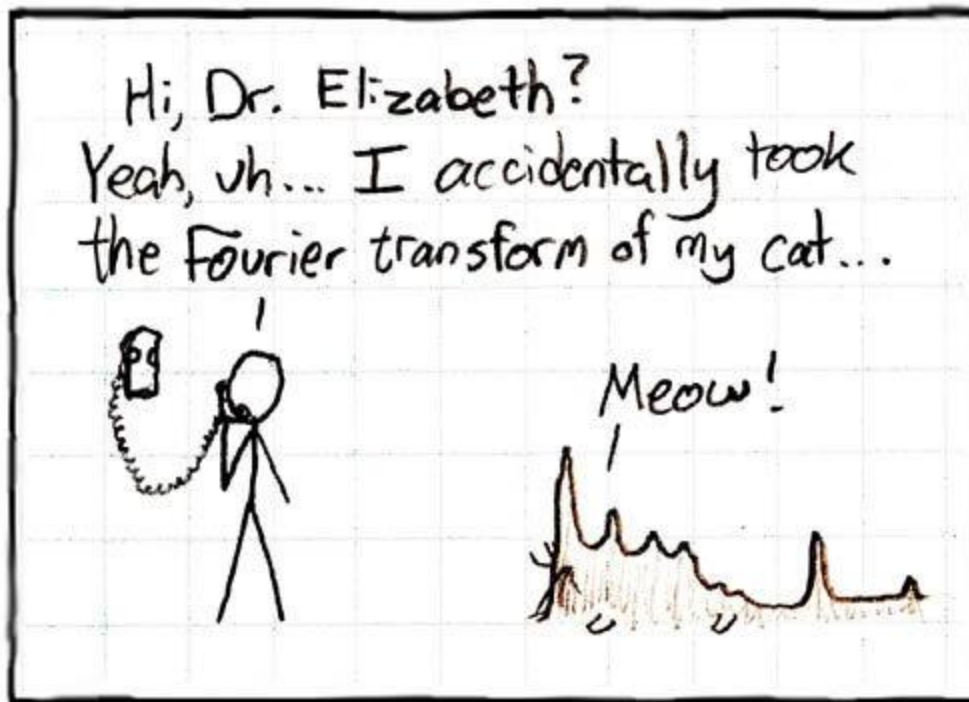
- We think of music in terms of frequencies at different magnitudes





# Other signals

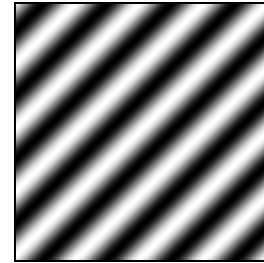
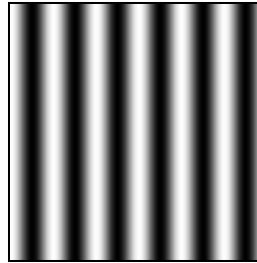
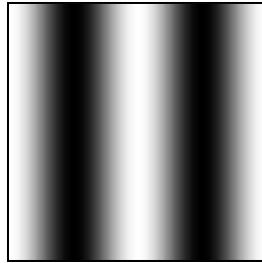
- We can also think of all kinds of other signals the same way



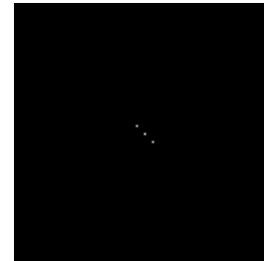
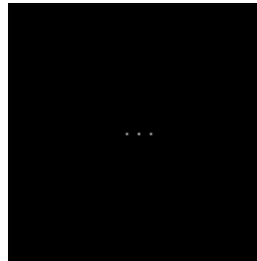
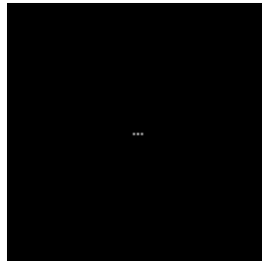
xkcd.com

# Fourier analysis in images

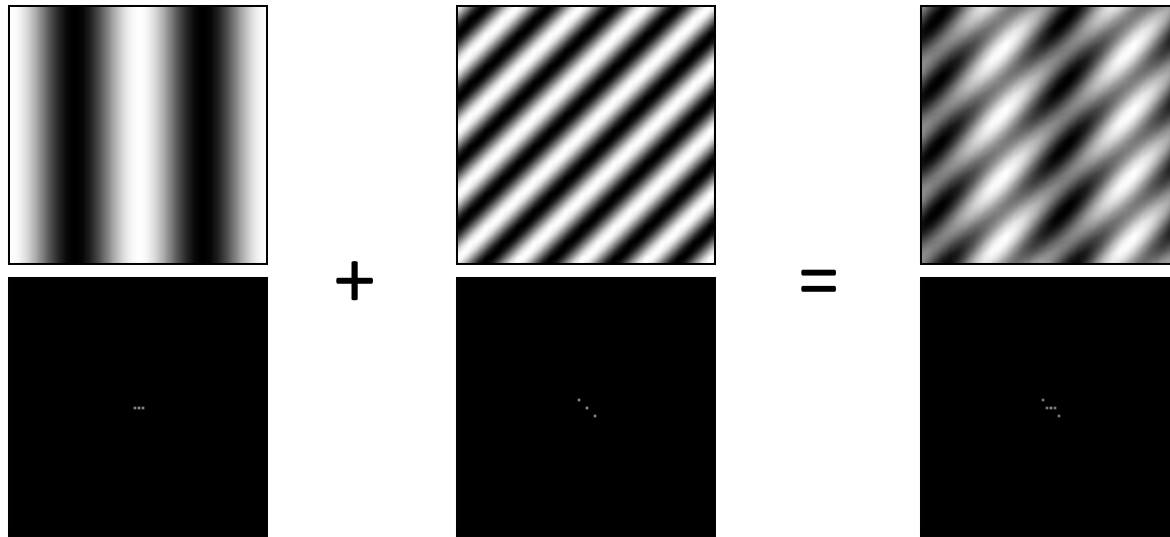
Intensity Image



Fourier Image



# Signals can be composed



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

# 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:  $\phi = \tan^{-1} \frac{I(\omega)}{R(\omega)}$

# 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!

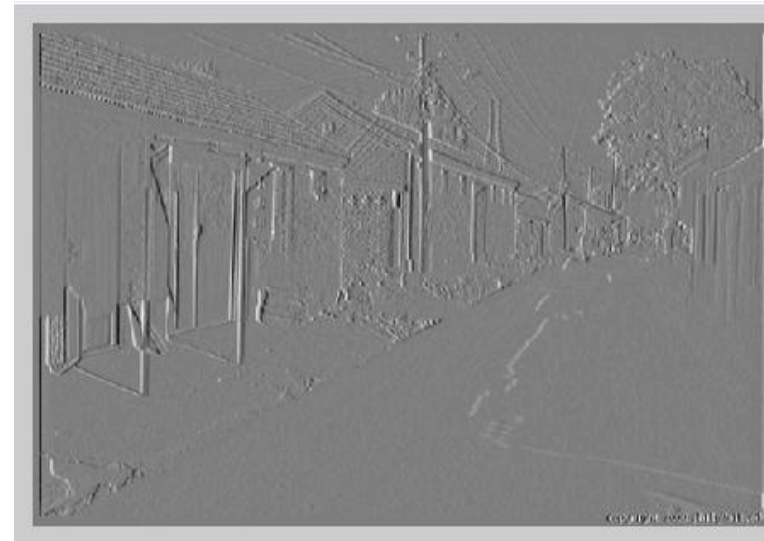
# Properties of Fourier Transforms

- 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

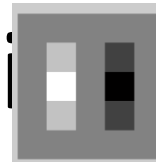
# Filtering in spatial dom

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

intensity image



# Filtering in frequency domain



FFT



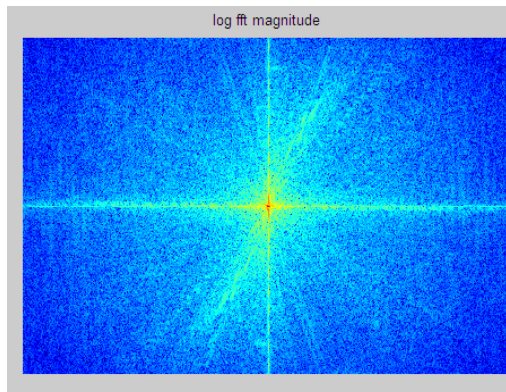
intensity image



FFT

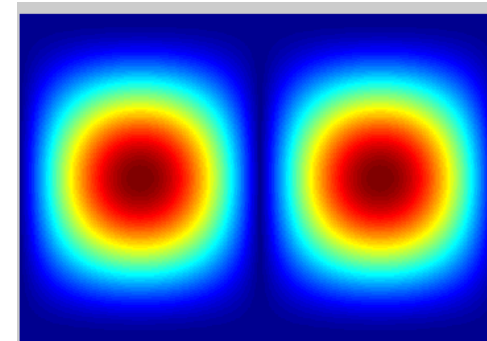


log fft magnitude

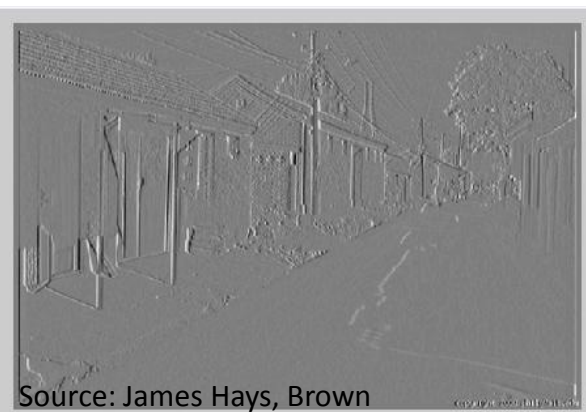
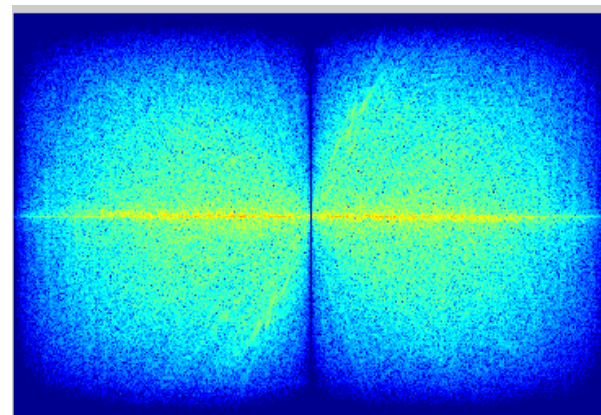


$\times$

$=$



Inverse FFT

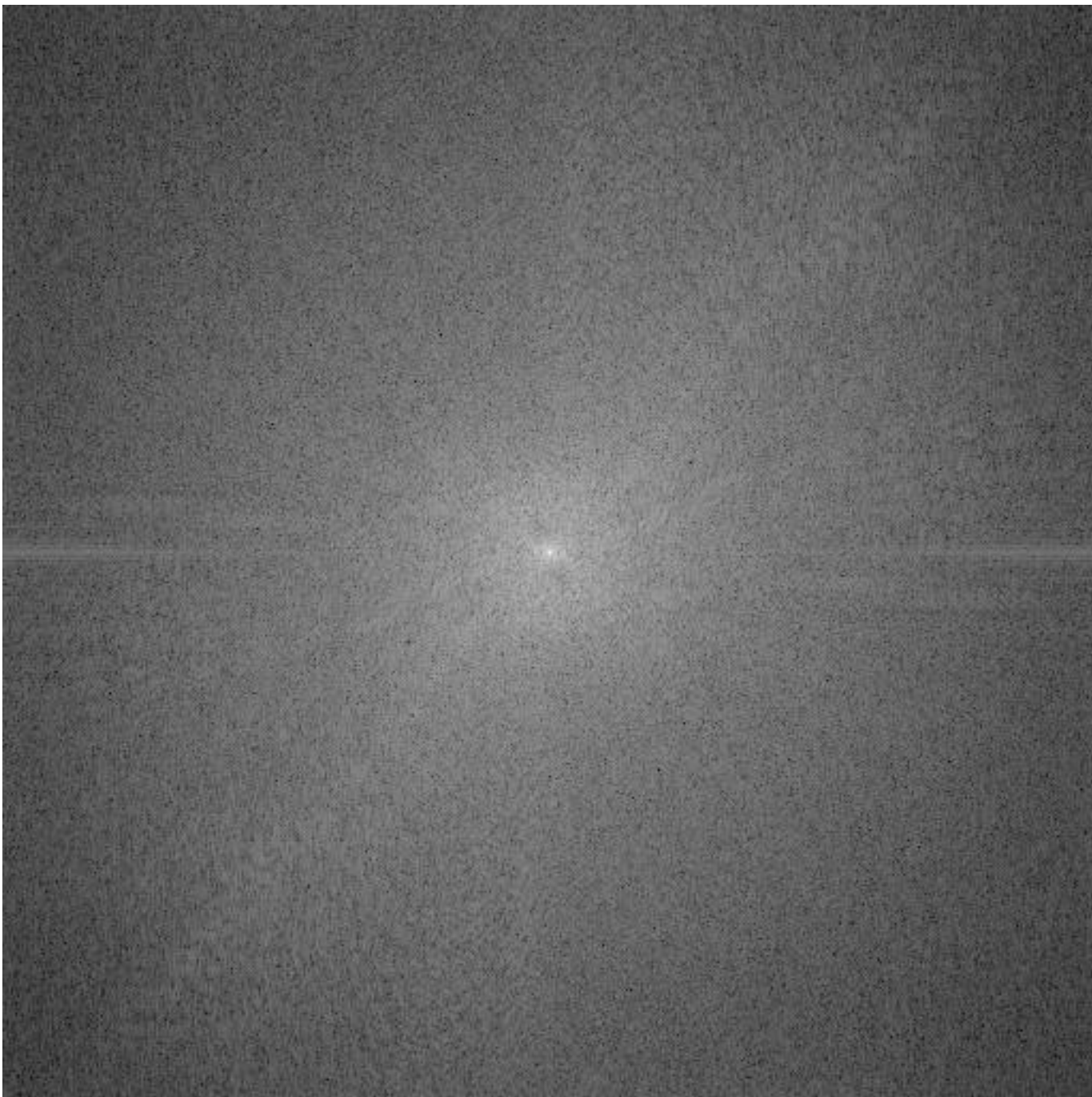




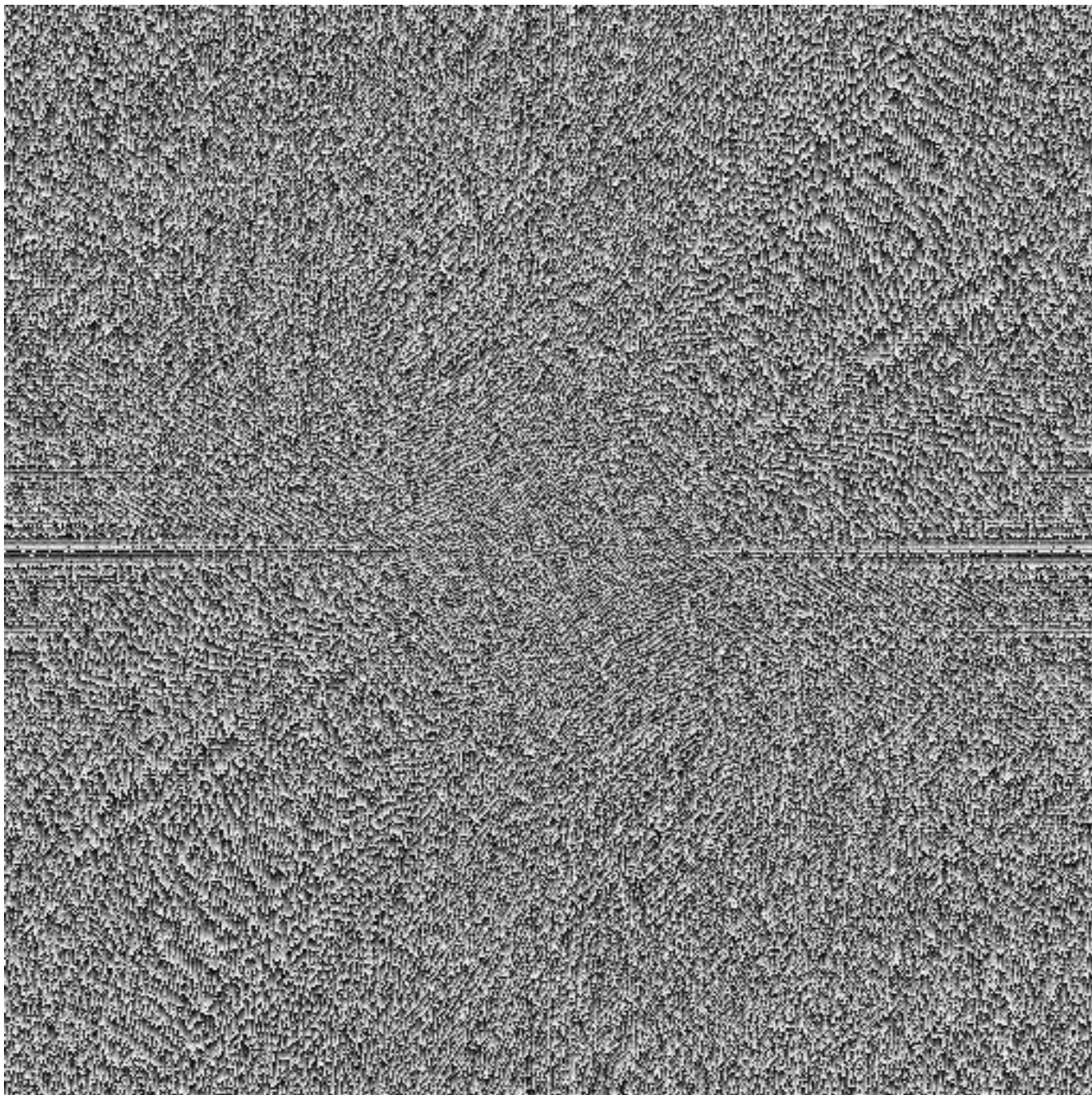


Source: Torralba, MIT

This is the  
magnitude  
transform  
of the  
cheetah pic



This is the  
phase  
transform  
of the  
cheetah pic

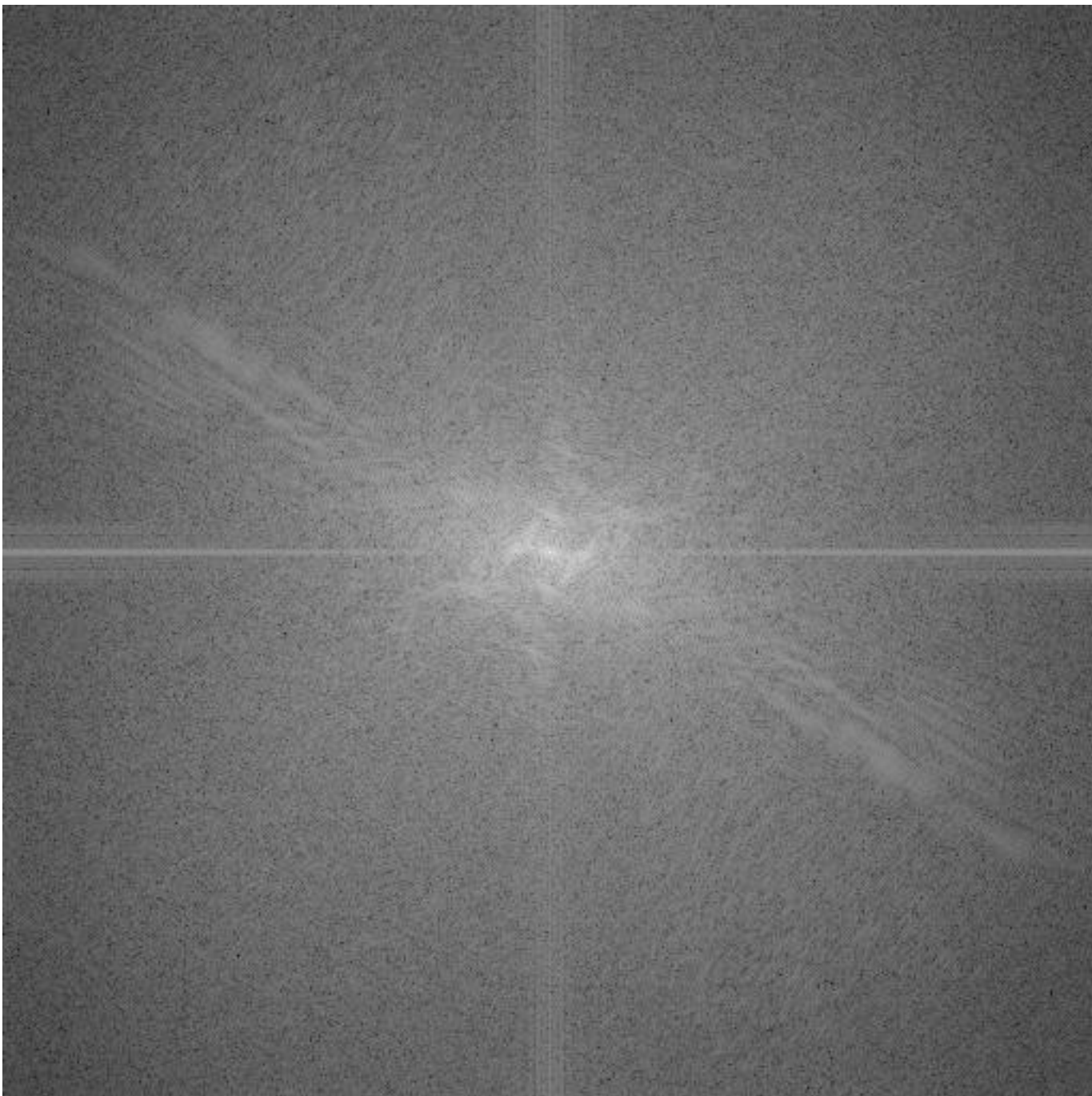




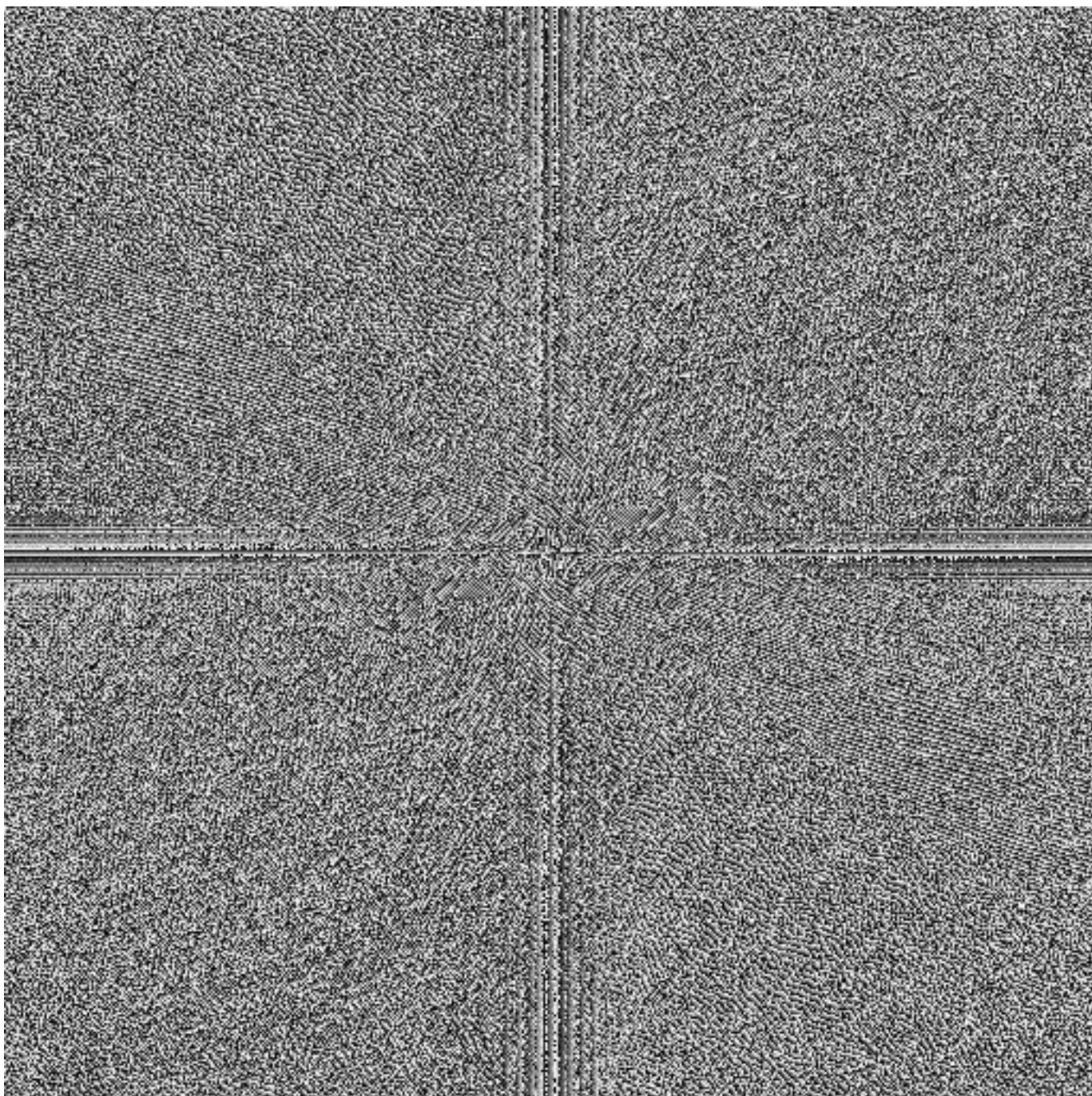
Source: Torralba, MIT



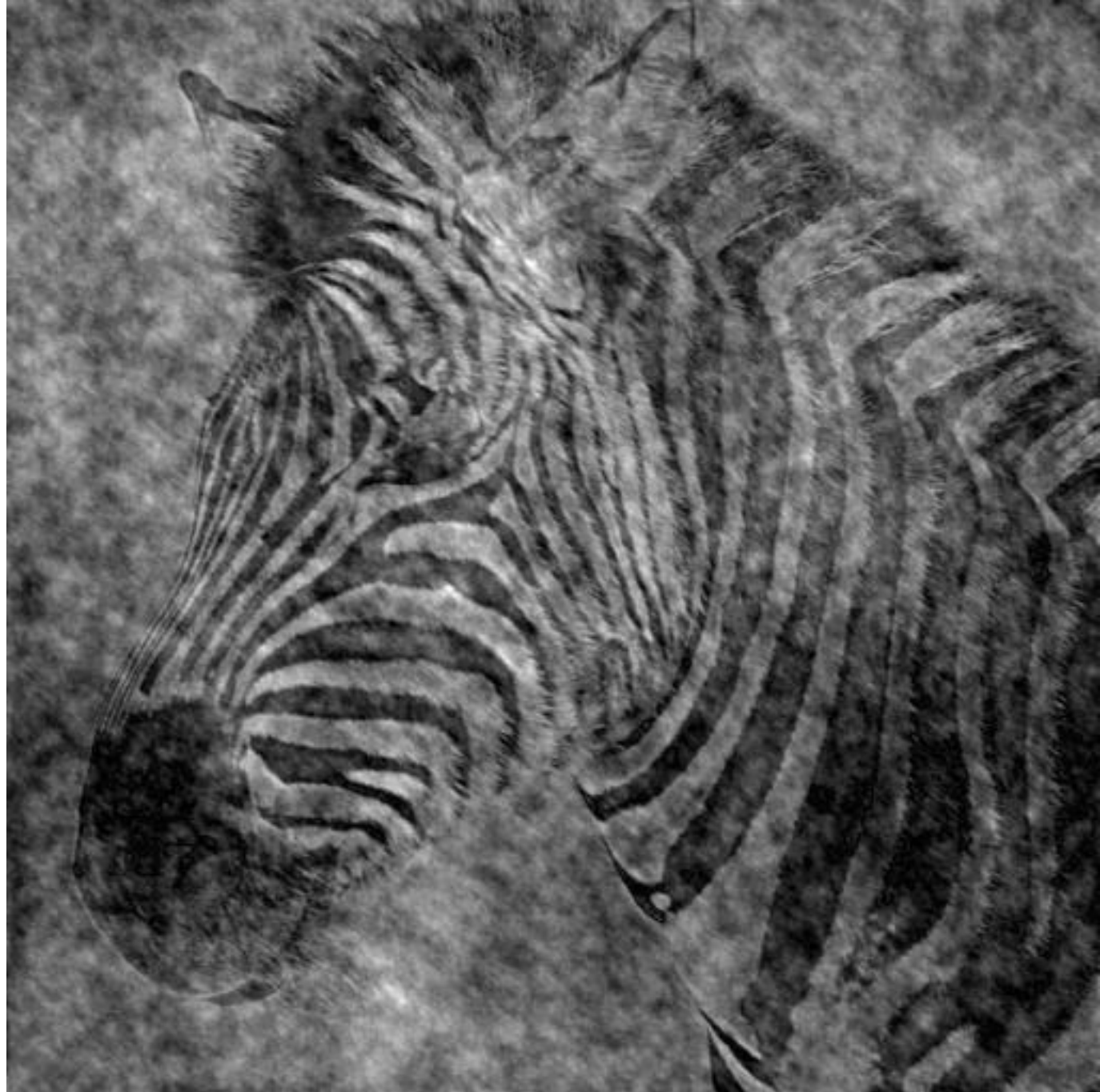
This is the  
magnitude  
transform  
of the zebra  
pic



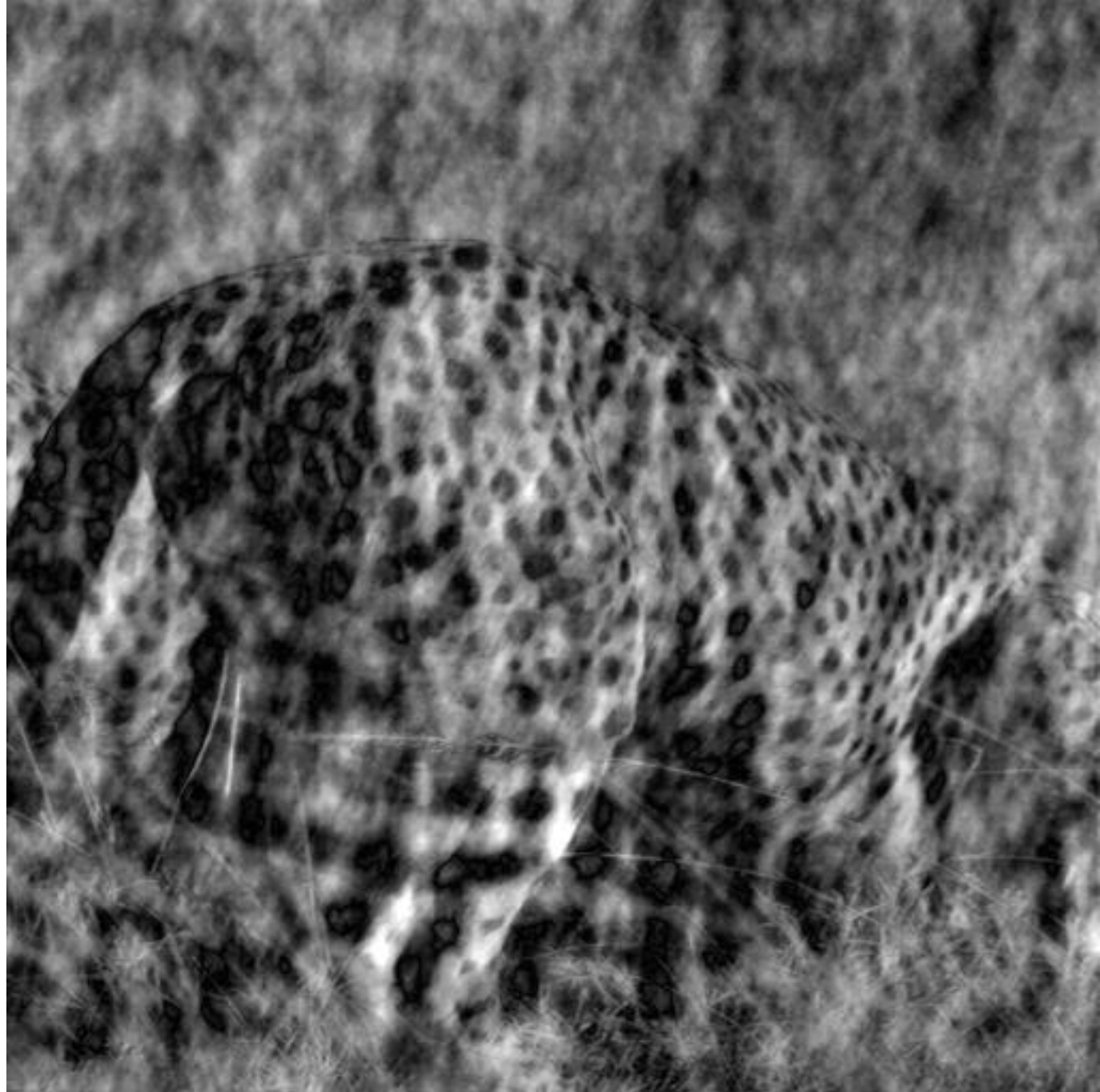
This is the  
phase  
transform  
of the zebra  
pic



Reconstruction  
with zebra  
phase, cheetah  
magnitude



Reconstruction  
with cheetah  
phase, zebra  
magnitude





# Phase and Magnitude

Image with cheetah phase  
(and zebra magnitude)

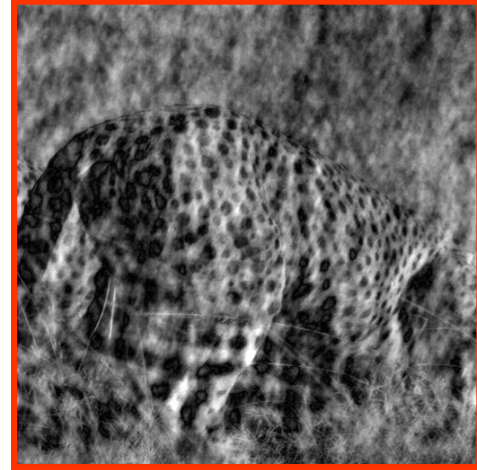
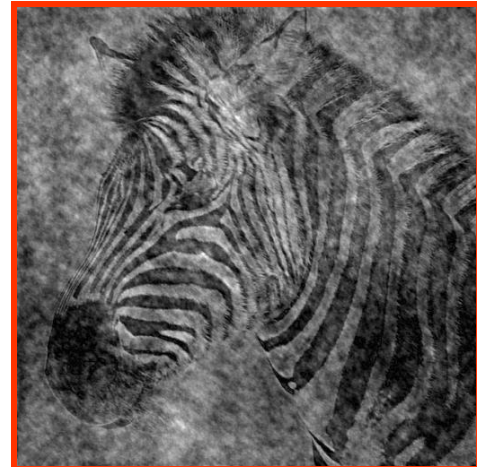


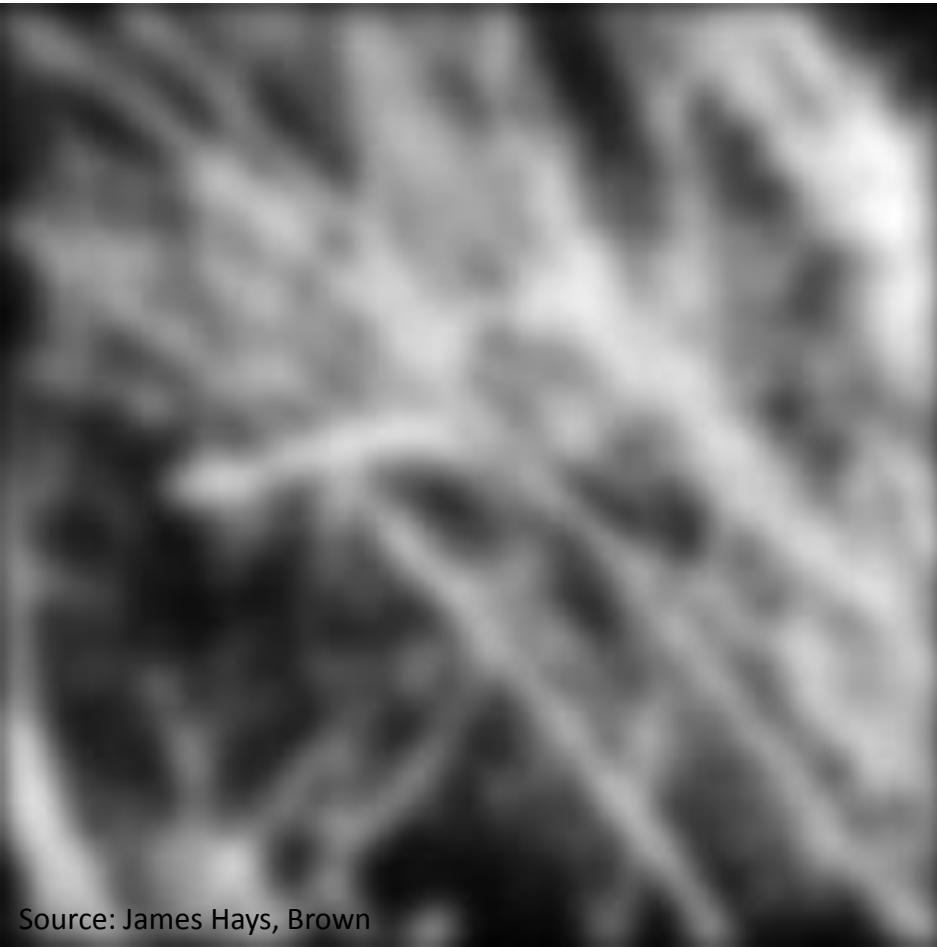
Image with zebra phase  
(and cheetah magnitude)



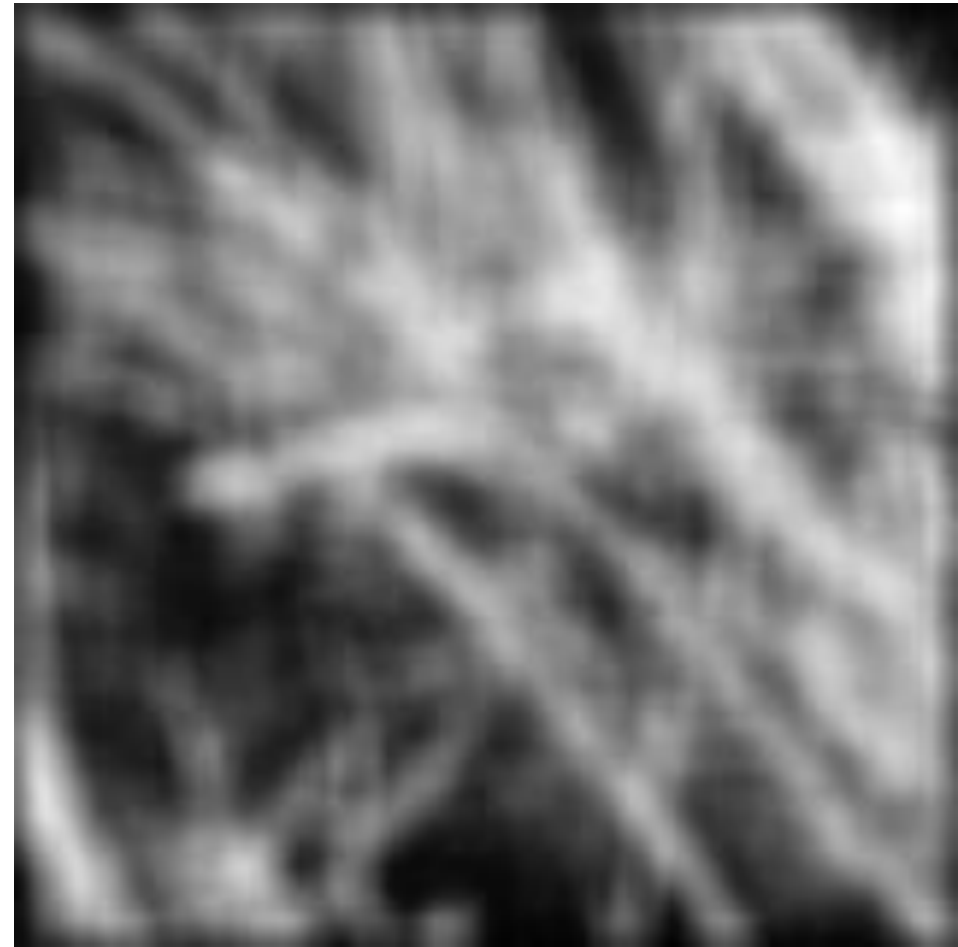
# Filtering

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

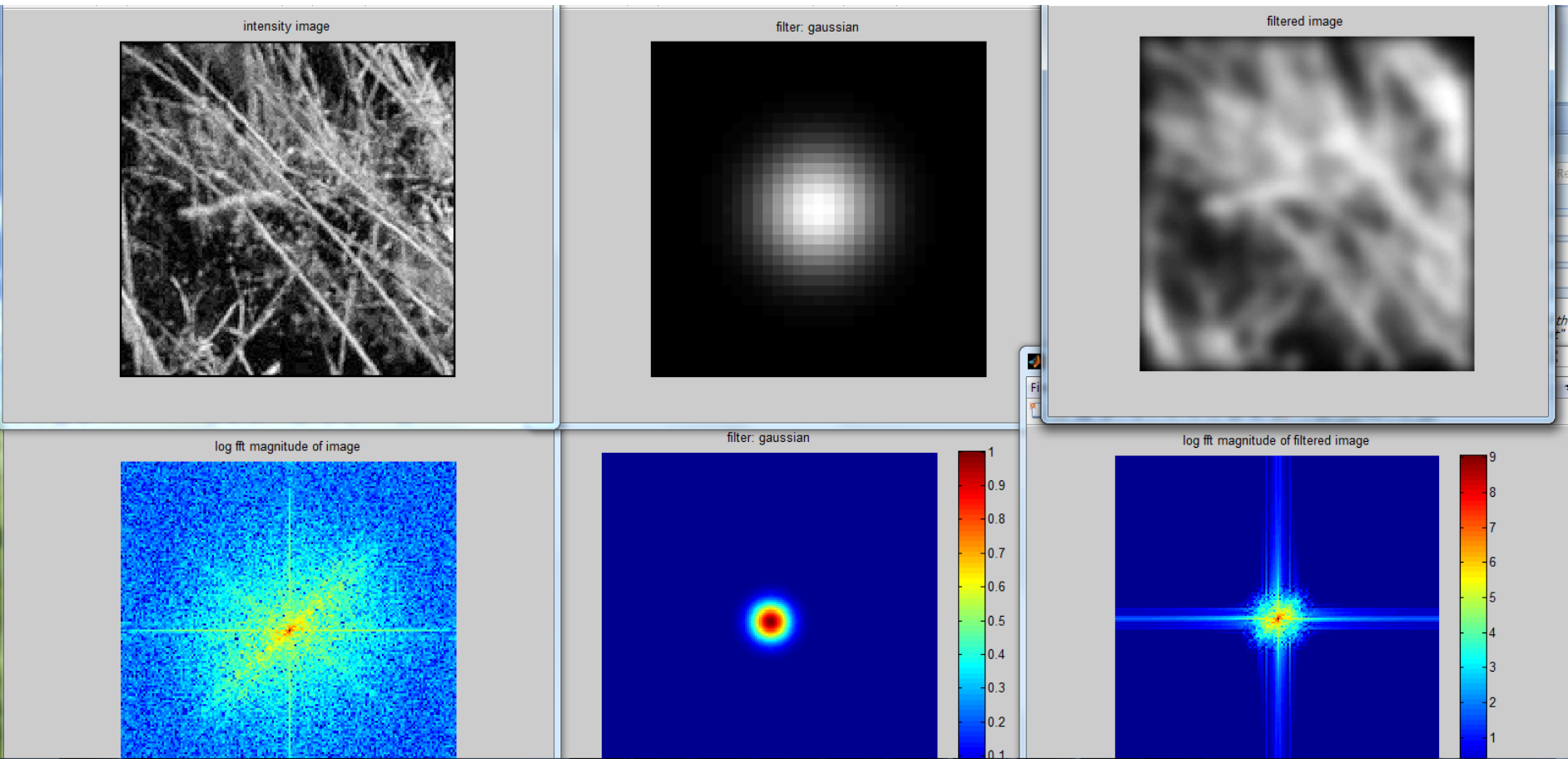
Gaussian



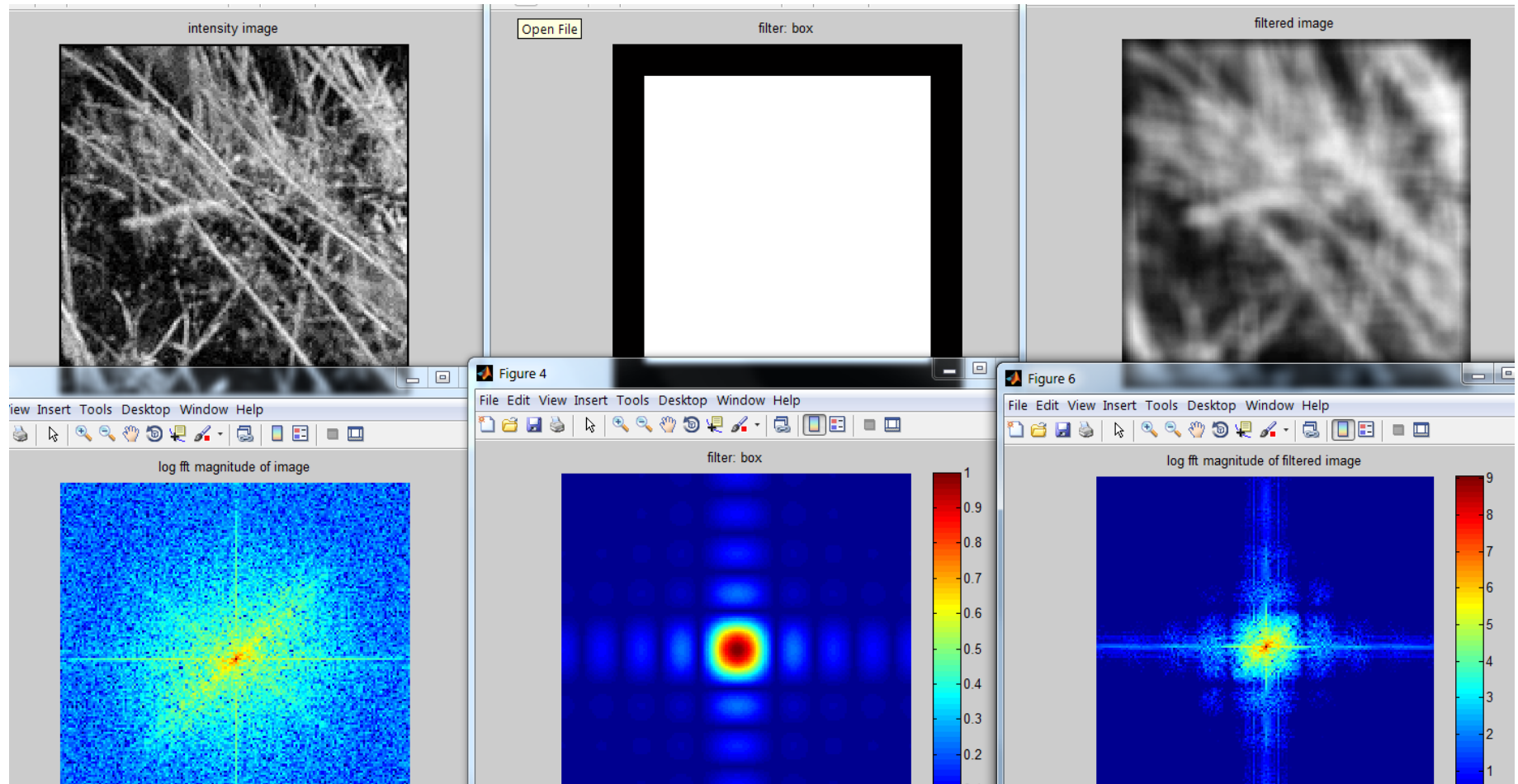
Box filter



# Gaussian



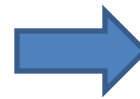
# Box Filter



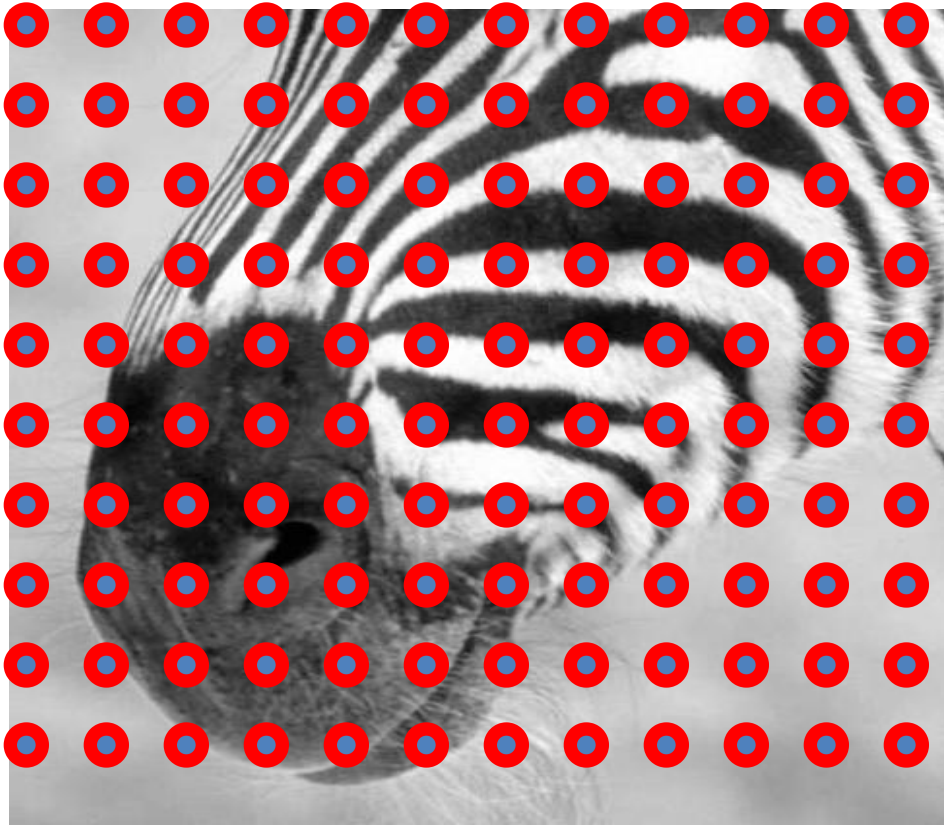


# Sampling

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



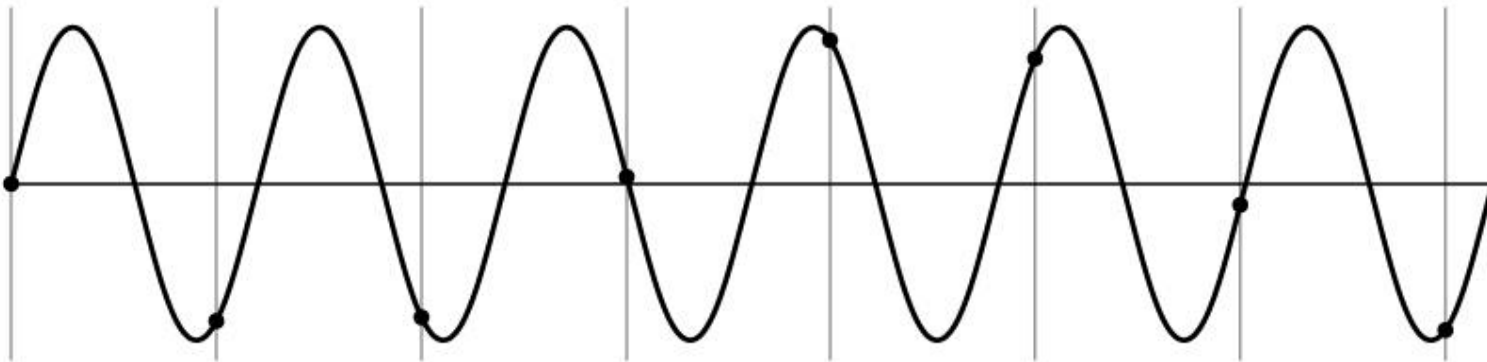
# Subsampling by a factor of 2



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

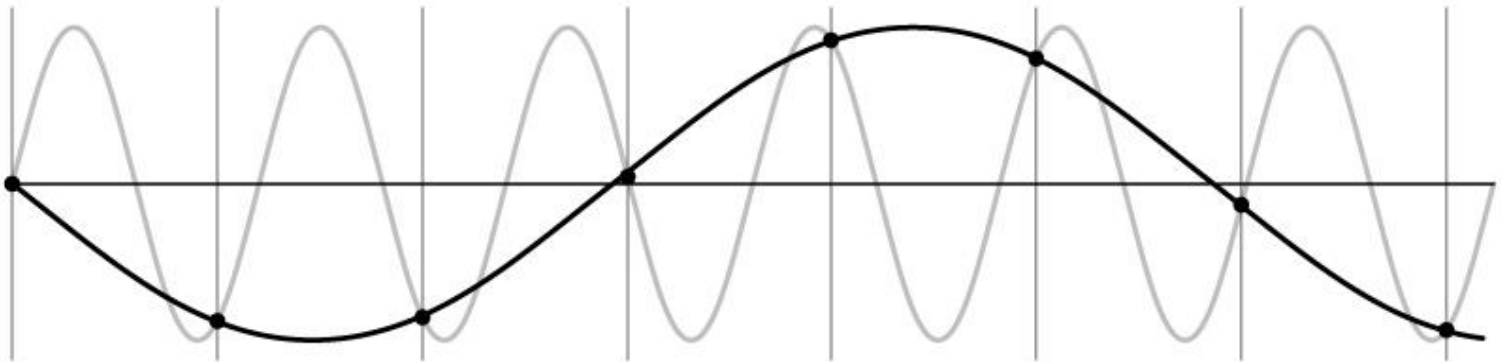
# Aliasing problem

- 1D example (sinewave):



# Aliasing problem

- 1D example (sinewave):





# Aliasing problem

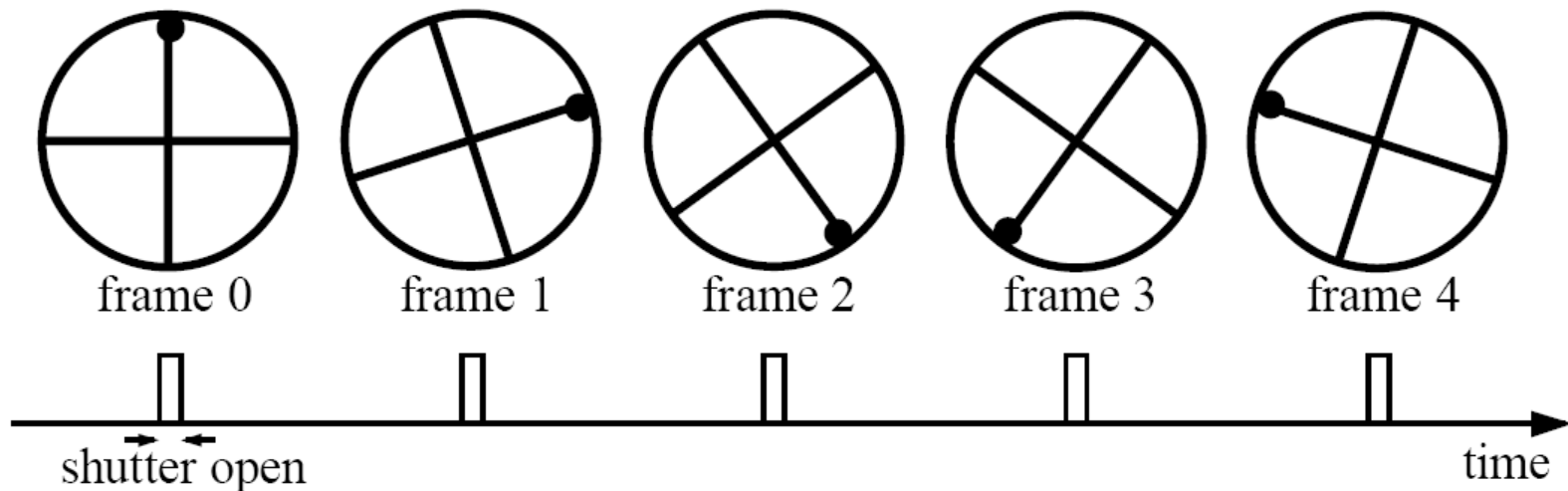
- Sub-sampling may be dangerous....
- Characteristic errors may appear:
  - “Wagon wheels rolling the wrong way in movies”
  - “Checkerboards disintegrate in ray tracing”
  - “Striped shirts look funny on color television”

# 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):

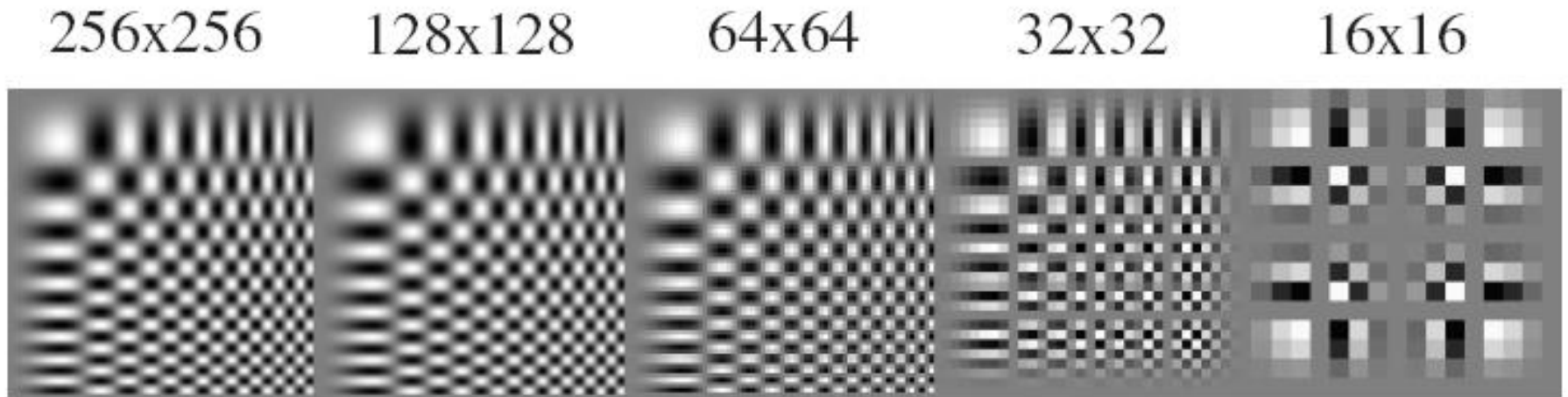


Without dot, wheel appears to be rotating slowly backwards!  
(counterclockwise)

# Aliasing in graphics

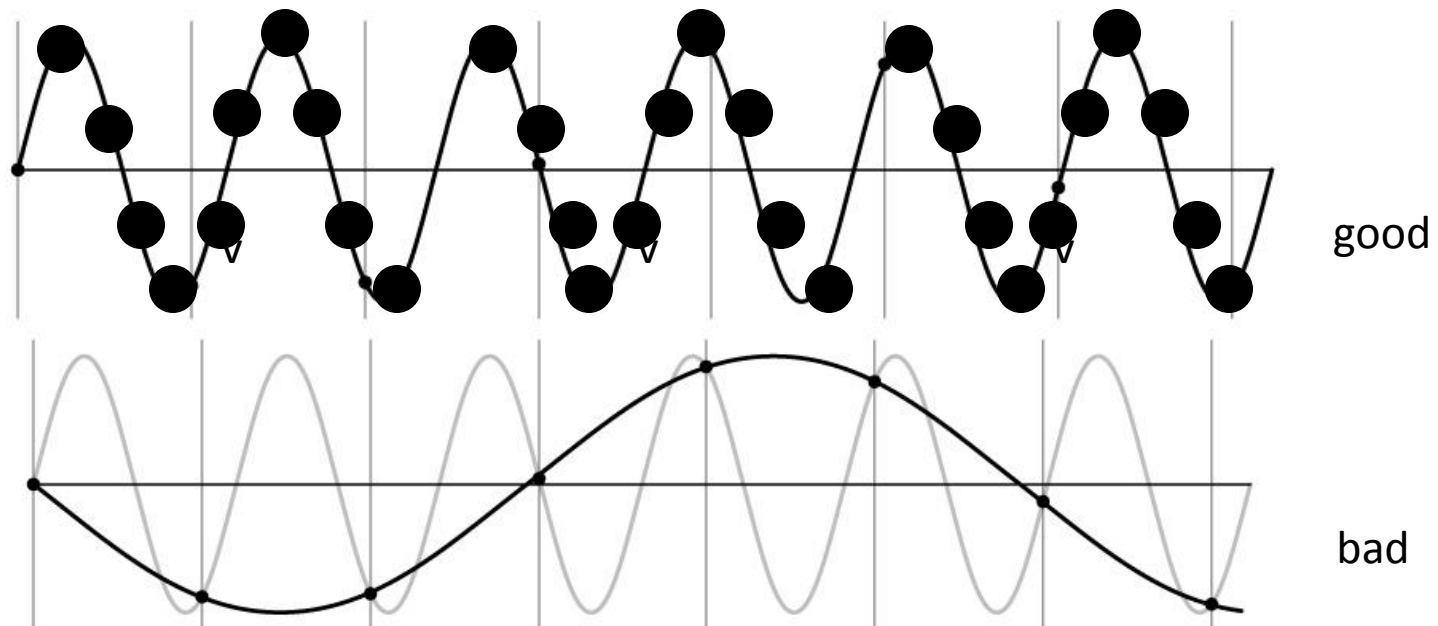


# Sampling and aliasing



# Nyquist-Shannon Sampling Theorem

- When sampling a signal at discrete intervals, the sampling frequency must be  $\geq 2 \times f_{\max}$
- $f_{\max}$  = max frequency of the input signal
- This will allow 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

# Algorithm for downsampling by factor of 2

1. Start with image(h, w)
2. Apply low-pass filter  
`im_blur = imfilter(image, fspecial('gaussian', 7, 1))`
3. Sample every other pixel  
`im_small = im_blur(1:2:end, 1:2:end);`

# Anti-aliasing

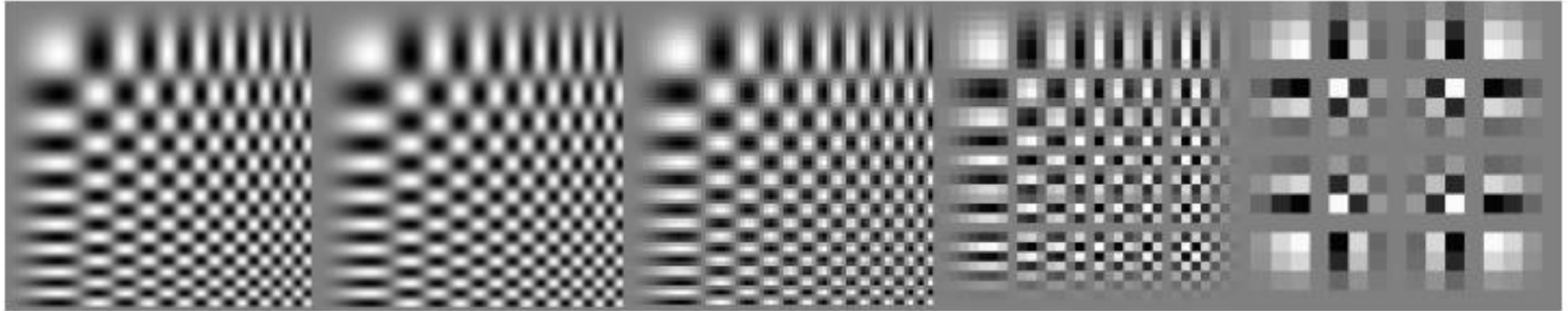
256x256

128x128

64x64

32x32

16x16



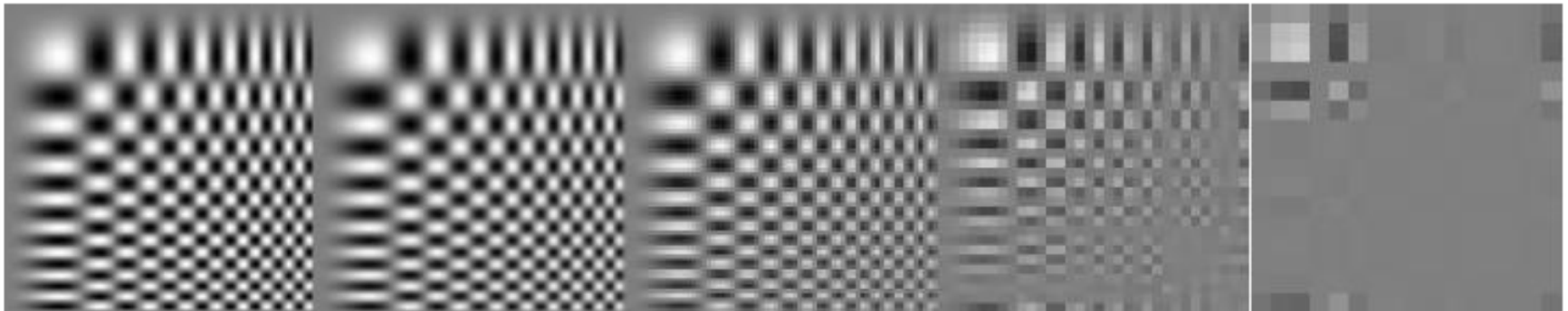
256x256

128x128

64x64

32x32

16x16





# Subsampling without pre-filtering



$1/2$



$1/4$  (2x zoom)



$1/8$  (4x zoom)

# Subsampling with Gaussian pre-filtering



Gaussian  $1/2$



G  $1/4$



G  $1/8$

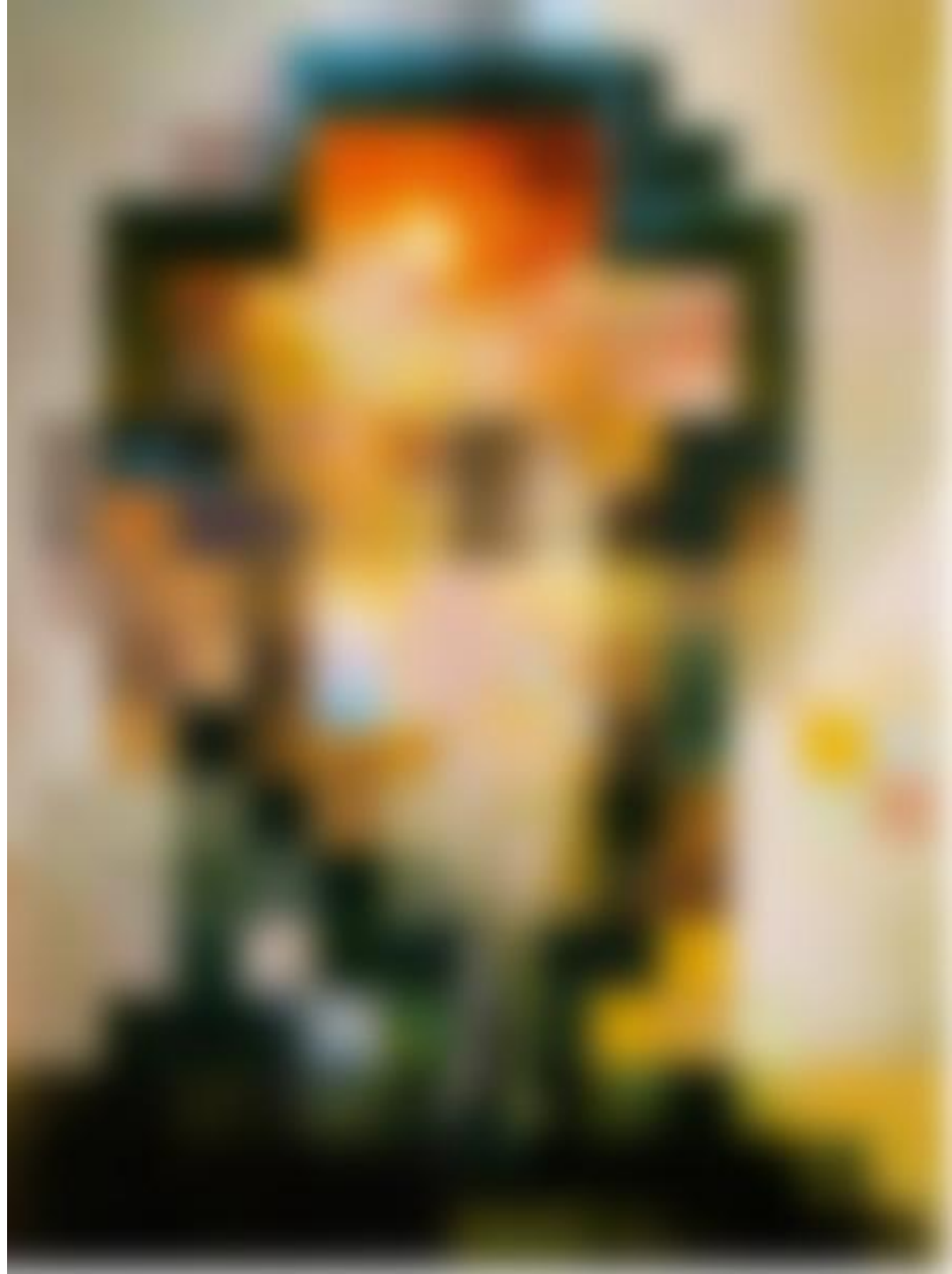


Salvador Dali invented Hybrid Images?

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

Source: James Hays, Brown



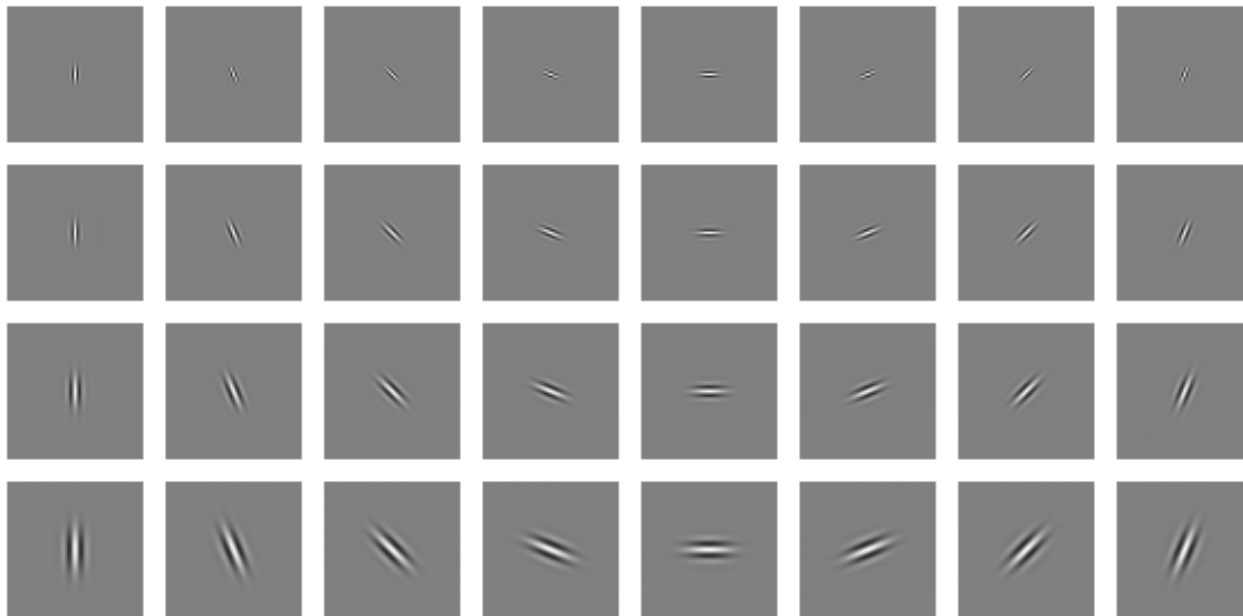


Source: James Hays, Brown



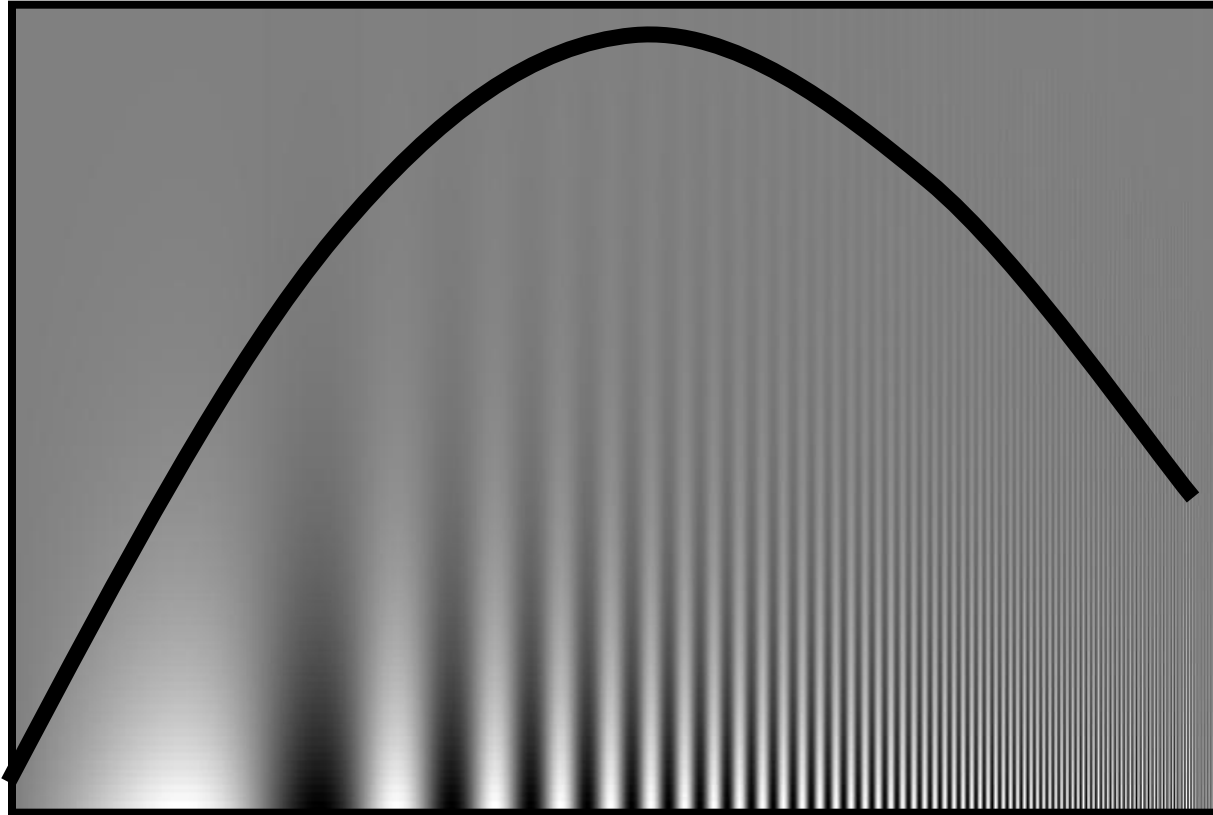
# Clues from Human Perception

- 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



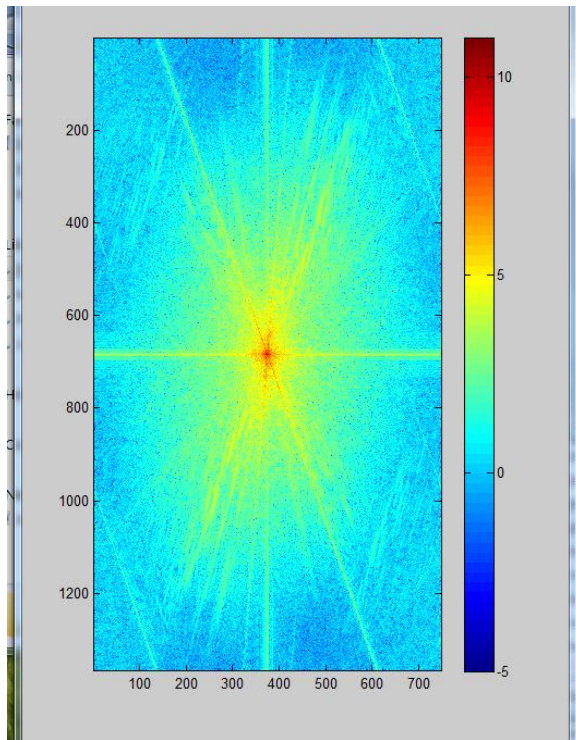
Early Visual Processing: Multi-scale edge and blob filters

# Campbell-Robson contrast sensitivity curve

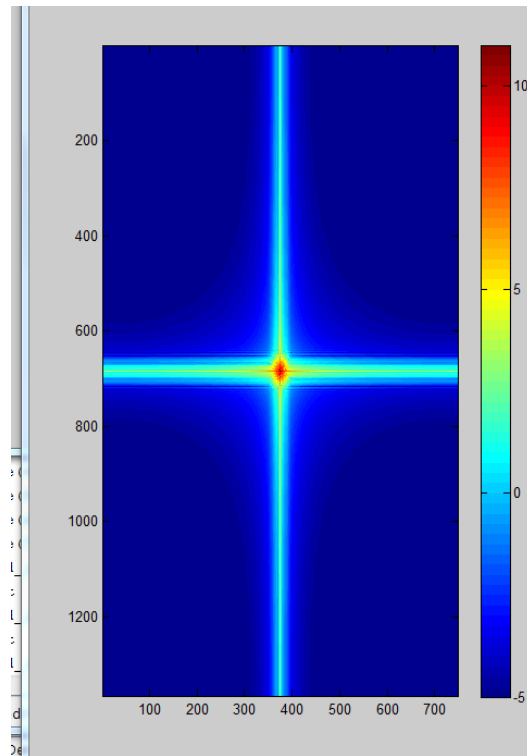


# Hybrid Image in FFT

Hybrid Image



Low-passed Image



High-passed Image

