



计算机视觉表征与识别

Chapter 3: Frequency Domain and Sampling

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媒体计算课题组

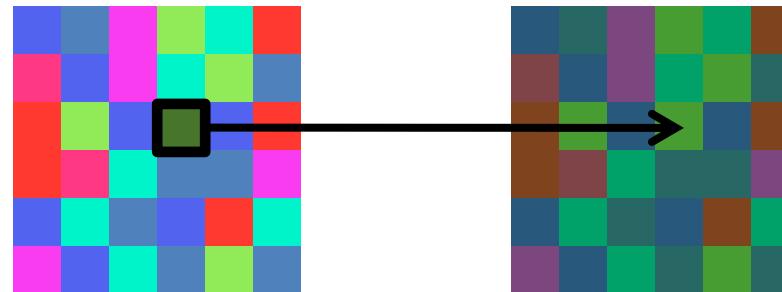
<http://mcg.nju.edu.cn/>



Image filtering

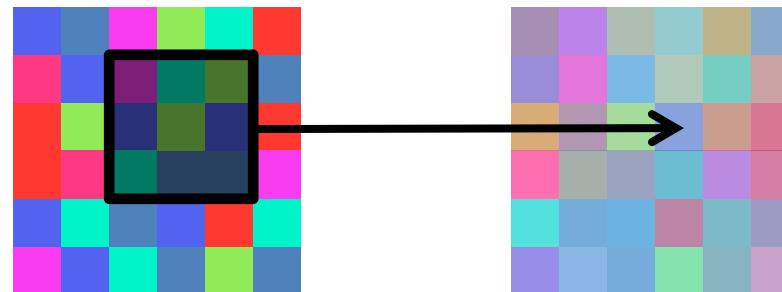


Point Operation



point processing

Neighborhood Operation



“filtering”



Image filtering



- Compute a function of the local neighborhood at each pixel in the image
 - Function specified by a “filter” or mask saying how to combine values from neighbors.
- Uses of filtering:
 - Enhance an image (denoise, resize, etc)
 - Extract information (texture, edges, etc)
 - Detect patterns (template matching)

Adapted from Derek Hoiem



Three views of filtering



- **Image filters in spatial domain**
 - Filter is a mathematical operation on values of each patch
 - 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



Common types of noise

- **Salt and pepper noise:** random occurrences of black and white pixels
- **Impulse noise:** random occurrences of white pixels
- **Gaussian noise:** variations in intensity drawn from a Gaussian normal distribution



Original



Salt and pepper noise



Impulse noise



Gaussian noise



Correlation filtering



Say the averaging window size is $2k+1 \times 2k+1$:

$$G[i, j] = \underbrace{\frac{1}{(2k+1)^2}}_{\text{Attribute uniform weight to each pixel}} \sum_{u=-k}^k \underbrace{\sum_{v=-k}^k F[i+u, j+v]}_{\text{Loop over all pixels in neighborhood around image pixel } F[i,j]}$$

Now generalize to allow different weights depending on neighboring pixel's relative position:

$$G[i, j] = \sum_{u=-k}^k \sum_{v=-k}^k \underbrace{H[u, v]}_{\text{Non-uniform weights}} F[i+u, j+v]$$



Averaging filter

- What values belong in the kernel H for the moving average example?

$F[x, y]$									
0	0	0	0	0	0	0	0	0	0
0	0	0	0	0	0	0	0	0	0
0	0	0	90	90	90	90	90	0	0
0	0	0	90	90	90	90	90	0	0
0	0	0	90	90	90	90	90	0	0
0	0	0	90	90	90	90	90	0	0
0	0	0	90	0	90	90	90	0	0
0	0	0	90	90	90	90	90	0	0
0	0	0	0	0	0	0	0	0	0
0	0	90	0	0	0	0	0	0	0
0	0	0	0	0	0	0	0	0	0



$G[x, y]$									
	0	20	40	60	60				

$$\frac{1}{9} \begin{matrix} 1 & 1 & 1 \\ 1 & 1 & ? \\ 1 & 1 & 1 \end{matrix}$$

“box filter”

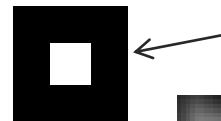
$$G = H \otimes F$$



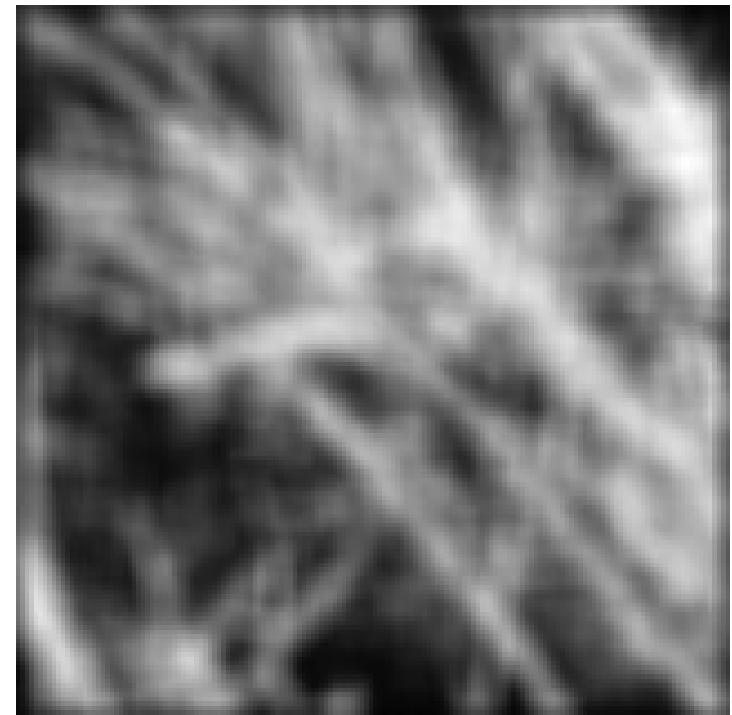
Smoothing by averaging



original



depicts box filter:
white = high value, black = low value



filtered

What if the filter size was 5×5 instead of 3×3 ?



Gaussian filter

- What if we want nearest neighboring pixels to have the most influence on the output?

0	0	0	0	0	0	0	0	0	0
0	0	0	0	0	0	0	0	0	0
0	0	0	90	90	90	90	90	0	0
0	0	0	90	90	90	90	90	0	0
0	0	0	90	90	90	90	90	0	0
0	0	0	90	90	90	90	90	0	0
0	0	0	90	0	90	90	90	0	0
0	0	0	90	90	90	90	90	0	0
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0	0	0	0	0	0	0	0	0	0

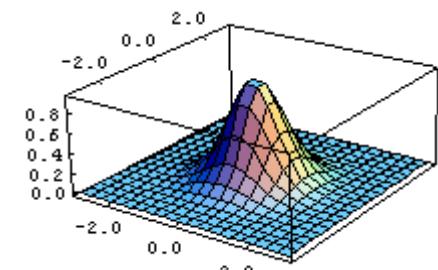
$F[x, y]$

$$\frac{1}{16} \begin{matrix} 1 & 2 & 1 \\ 2 & 4 & 2 \\ 1 & 2 & 1 \end{matrix}$$

$H[u, v]$

This kernel is an approximation of a 2d Gaussian function:

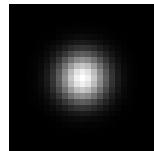
$$h(u, v) = \frac{1}{2\pi\sigma^2} e^{-\frac{u^2+v^2}{\sigma^2}}$$



- Removes high-frequency components from the image (“low-pass filter”).



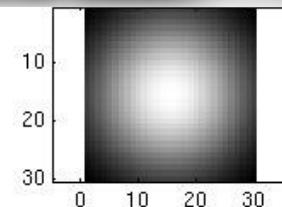
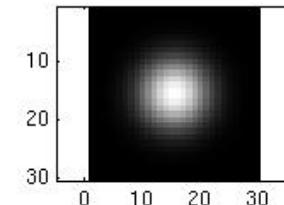
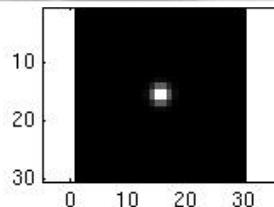
Smoothing with a Gaussian





Smoothing with a Gaussian

Parameter σ is the “scale” / “width” / “spread” of the Gaussian kernel, and controls the amount of smoothing.



```
for sigma=1:3:10
    h = fspecial('gaussian', fsize,
    sigma);
    out = imfilter(im, h);
    imshow(out);
    pause;
end
```



Convolution vs correlation



Definition of discrete 2D convolution:

$$(f * g)(x, y) = \sum_{i,j=-\infty}^{\infty} f(i, j)I(x - i, y - j)$$

↑ notice the flip

Definition of discrete 2D correlation:

$$(f * g)(x, y) = \sum_{i,j=-\infty}^{\infty} f(i, j)I(x + i, y + j)$$

↑ notice the lack of a flip

- Most of the time won't matter, because our kernels will be symmetric.



Separability



- In some cases, filter is separable, and we can factor into two steps:
 - Convolve all rows with a 1D filter
 - Convolve all columns with a 1D filter

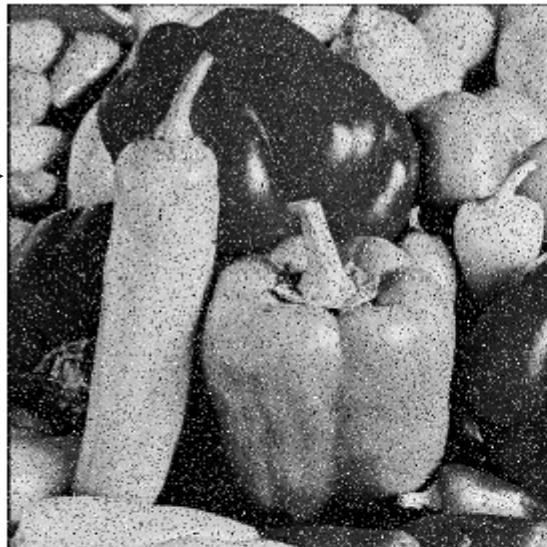
$$\begin{bmatrix} \frac{1}{9} & \frac{1}{9} & \frac{1}{9} \\ \frac{1}{9} & \frac{1}{9} & \frac{1}{9} \\ \frac{1}{9} & \frac{1}{9} & \frac{1}{9} \end{bmatrix} = \begin{bmatrix} 0 & \frac{1}{3} & 0 \\ 0 & \frac{1}{3} & 0 \\ 0 & \frac{1}{3} & 0 \end{bmatrix} \circ \begin{bmatrix} 0 & 0 & 0 \\ \frac{1}{3} & \frac{1}{3} & \frac{1}{3} \\ 0 & 0 & 0 \end{bmatrix}$$



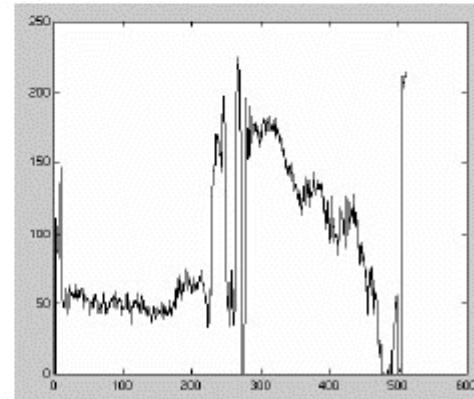
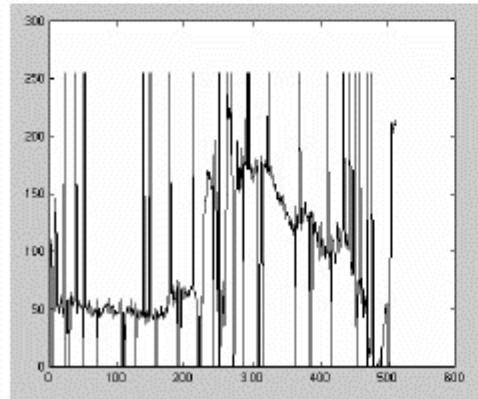
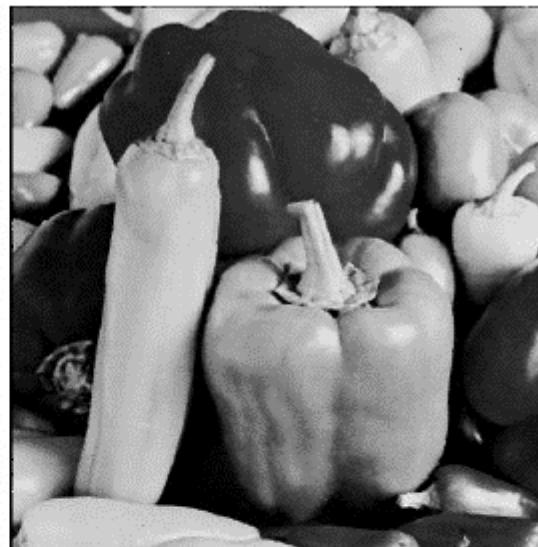
Median filter



Salt and
pepper
noise



Median
filtered



Plots of a row of the image

Matlab: output im = medfilt2(im, [h w]);



Objective of bilateral filtering



Smooth texture

Preserve edges



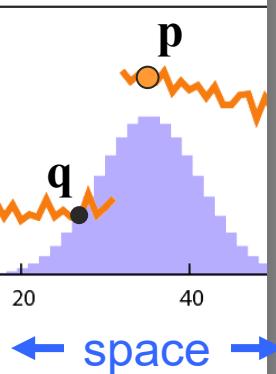
Definition

Gaussian blur

$$I_p^b = \sum_{q \in S} G_{\sigma_s}(\|p - q\|) I_q$$

space

- only spatial distance, intensity ignored

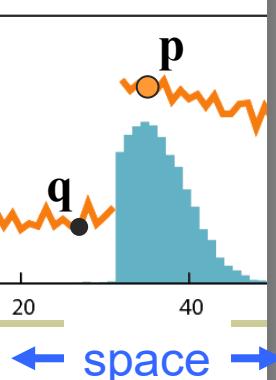


Bilateral filter

[Aurich 95, Smith 97, Tomasi 98]

$$I_p^{bf} = \frac{1}{W_p^{bf}} \sum_{q \in S} G_{\sigma_s}(\|p - q\|) G_{\sigma_r}(|I_p - I_q|) I_q$$

normalization



- spatial and range distances
- weights sum to 1

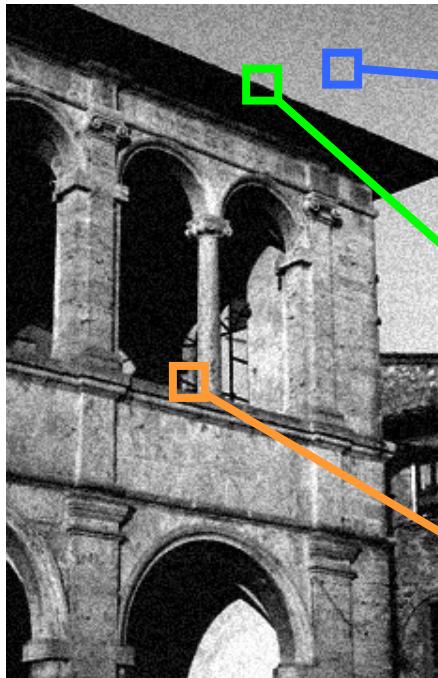


Example on a Real Image

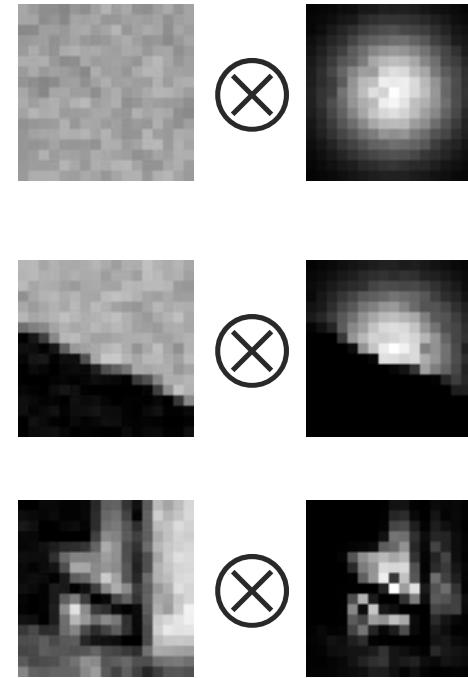


- Kernels can have complex, spatially varying shapes.

input



output





Three views of filtering



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 - Filter is a mathematical operation on values of each patch
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- **Image filters in the frequency domain**
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 - Denoising, sampling, image compression
- **Templates and Image Pyramids**
 - Filtering is a way to match a template to the image
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Today's Class

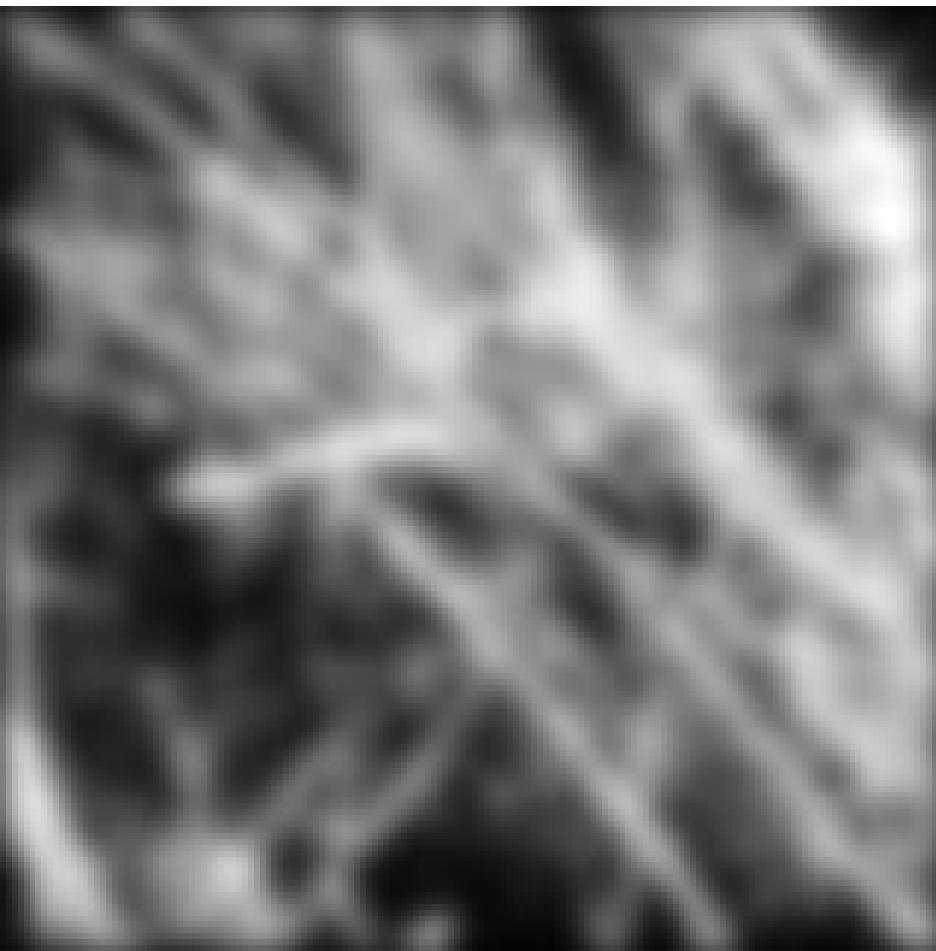


- Fourier transform and frequency domain
 - Frequency view of filtering
- Hybrid Image
- Sampling

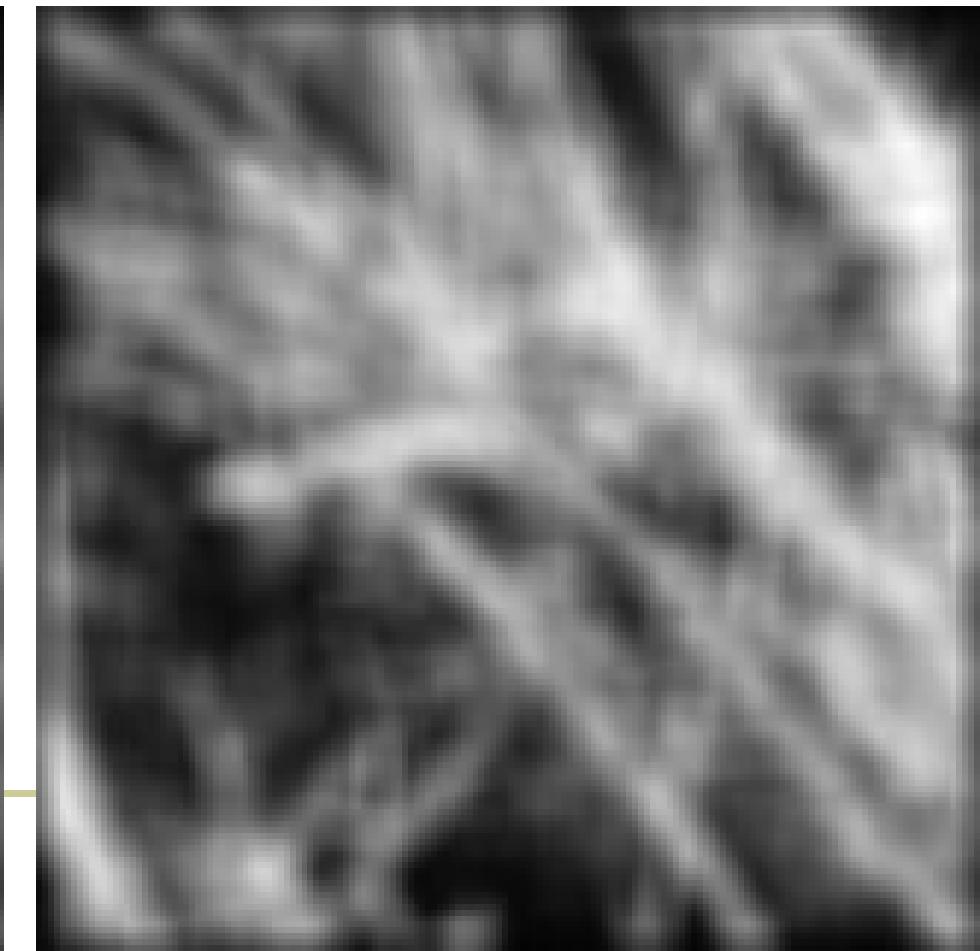


Why does the Gaussian give a nice smooth image, but the square filter give edgy artifacts?

Gaussian

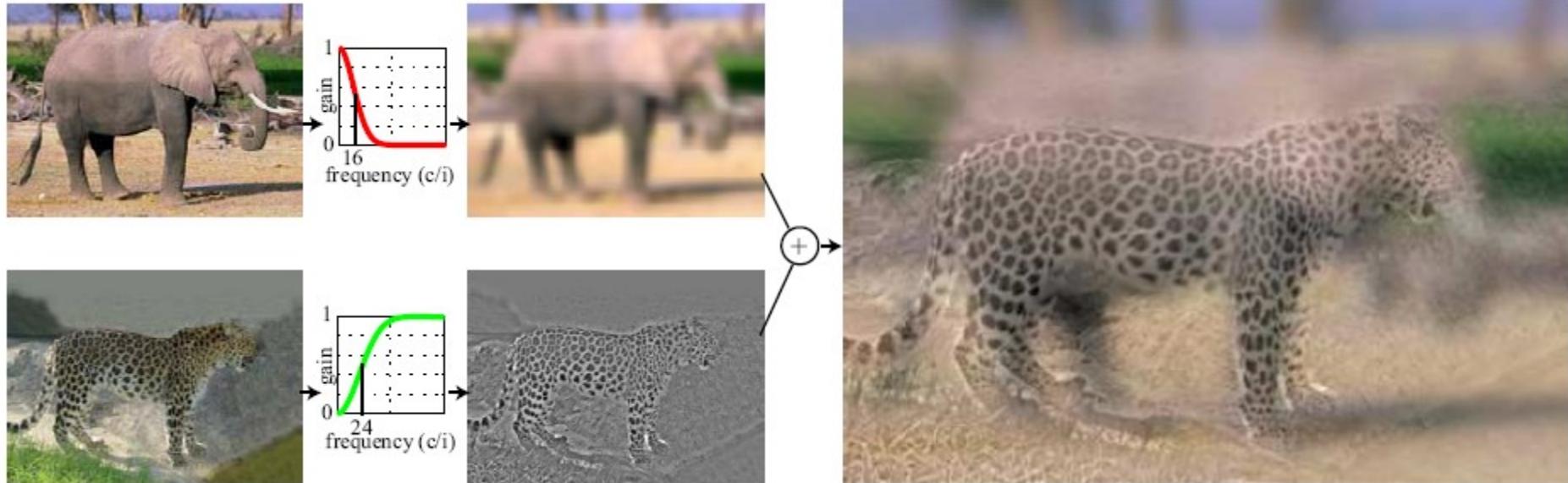


Box filter





Hybrid Images



- A. Oliva, A. Torralba, P.G. Schyns,
“Hybrid Images,” SIGGRAPH 2006





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





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





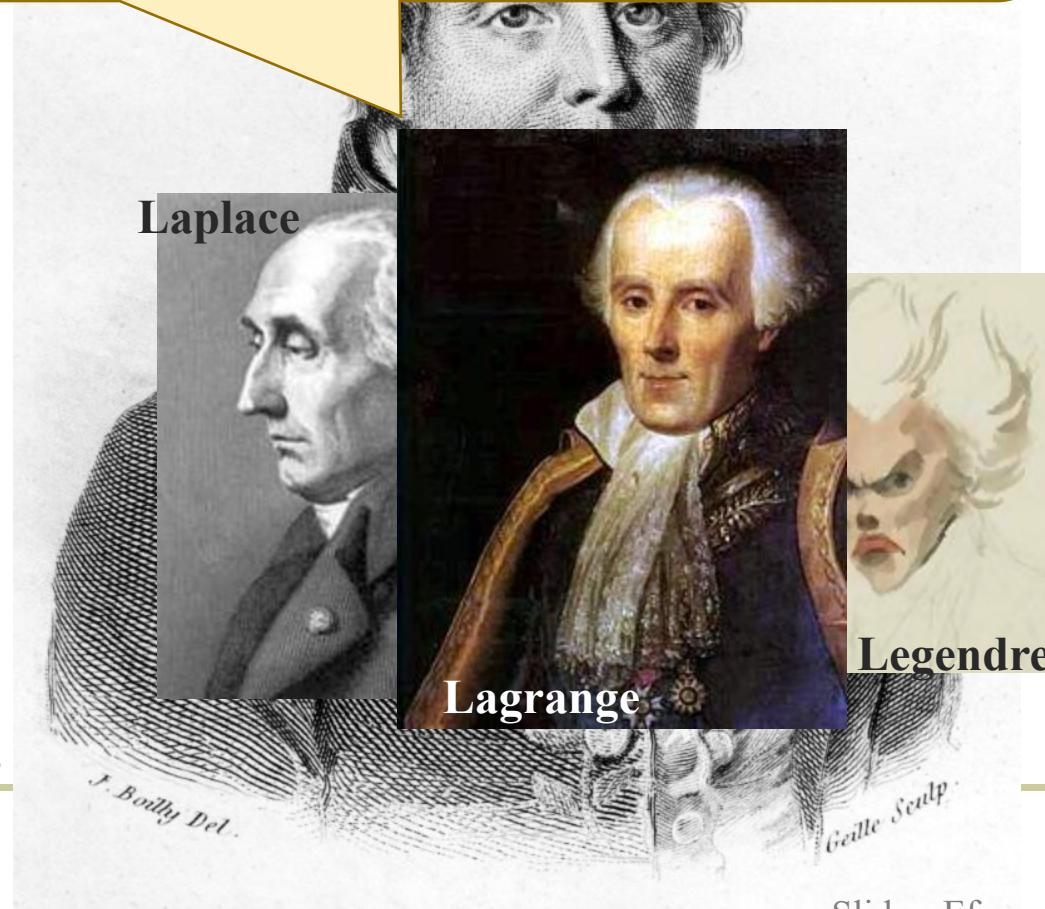
Jean Baptiste Joseph Fourier (1768-1830)

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





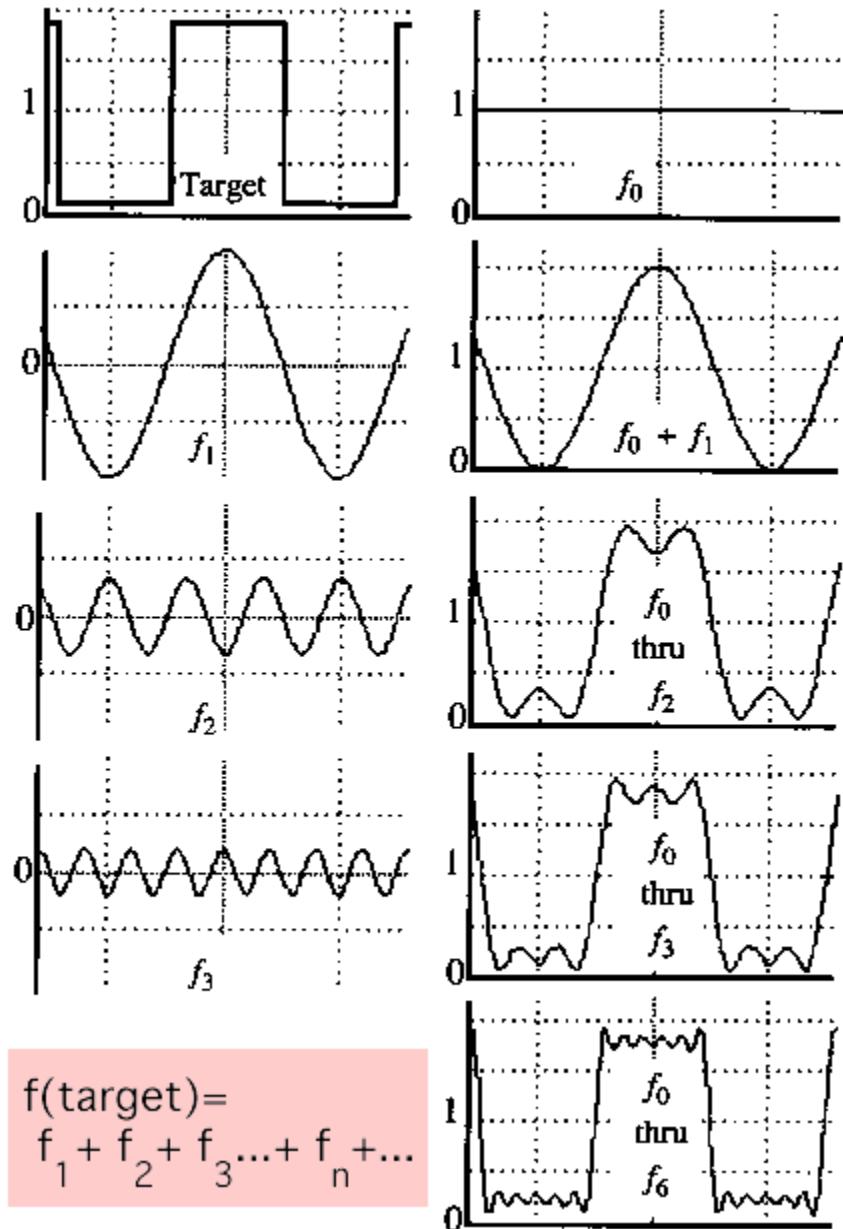
A sum of sines



Our building block:

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

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

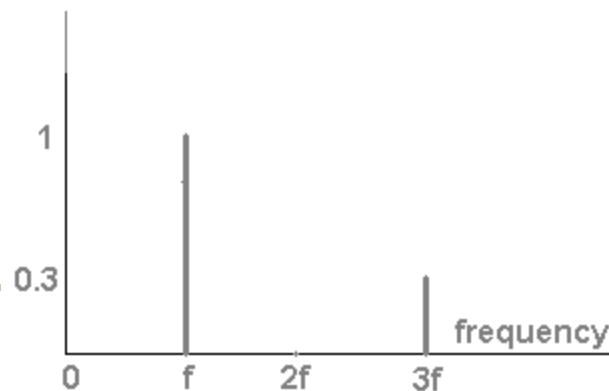
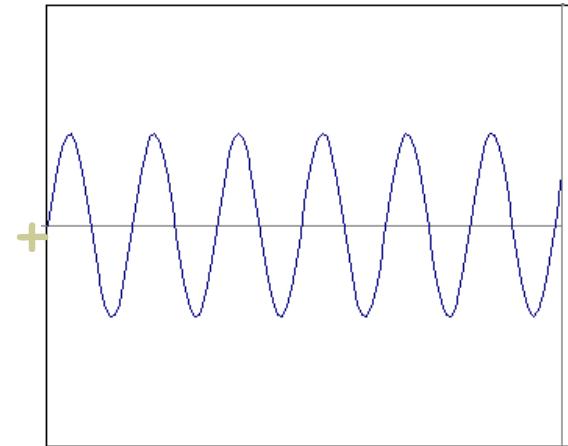
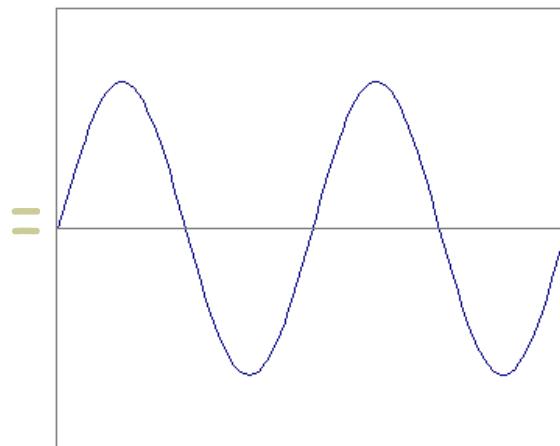
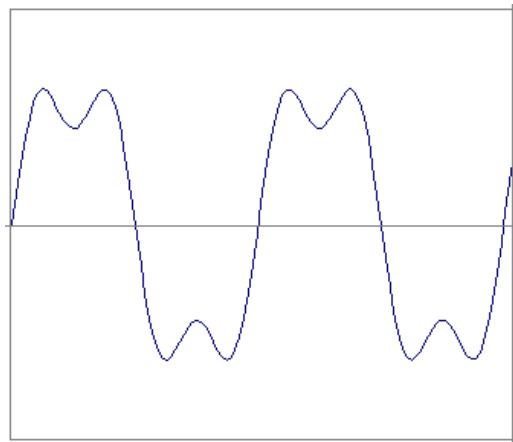




Frequency Spectra

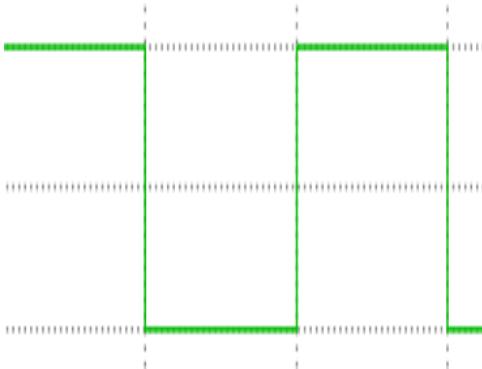


- example : $g(t) = \sin(2\pi f t) + (1/3)\sin(2\pi(3f)t)$



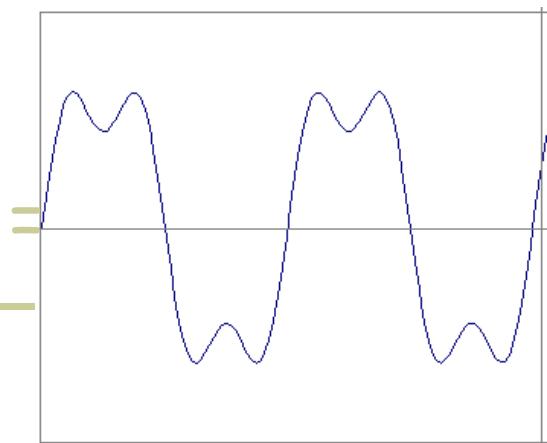
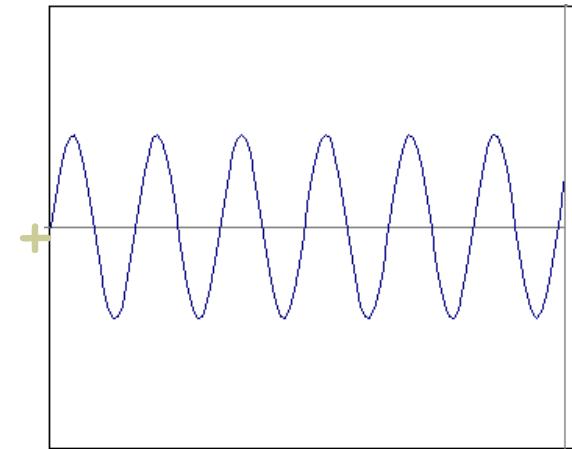
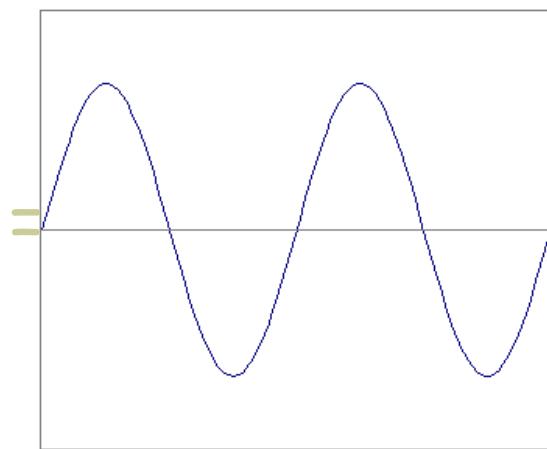
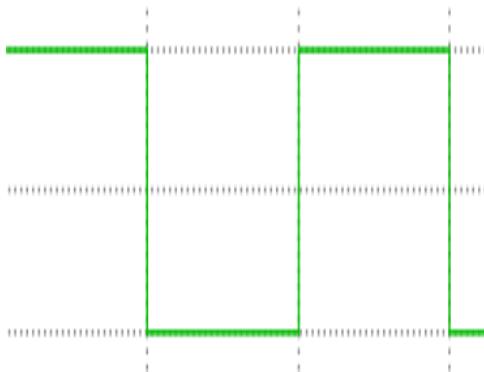


Frequency Spectra



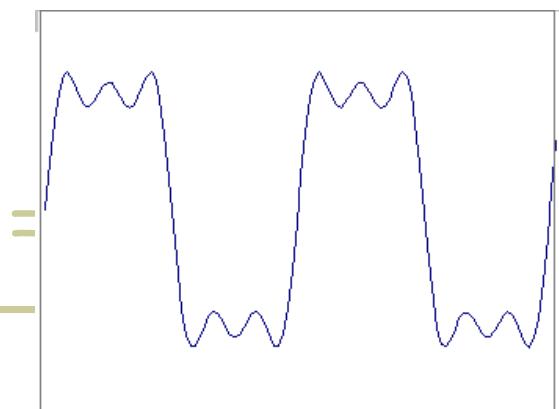
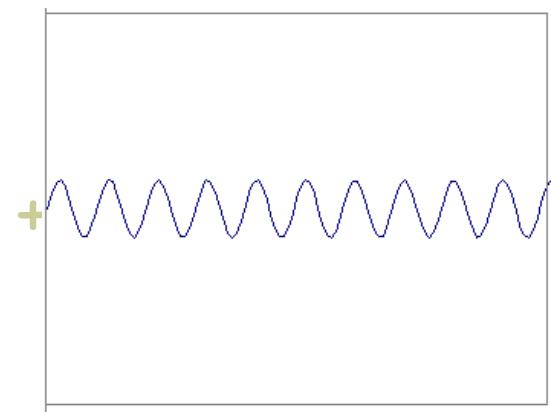
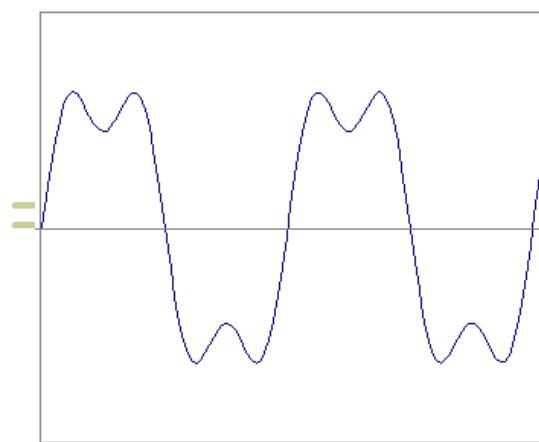
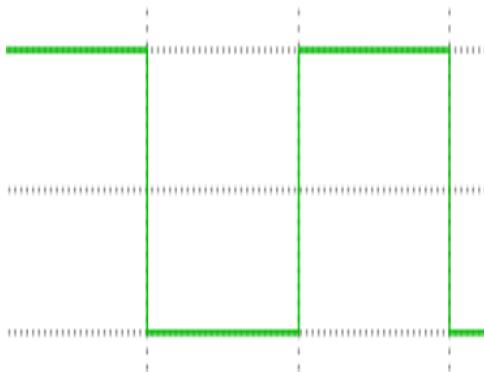


Frequency Spectra



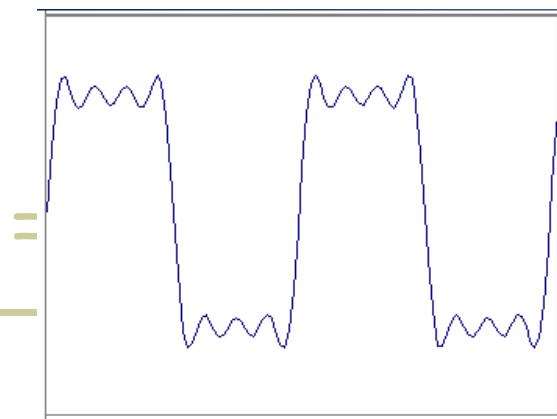
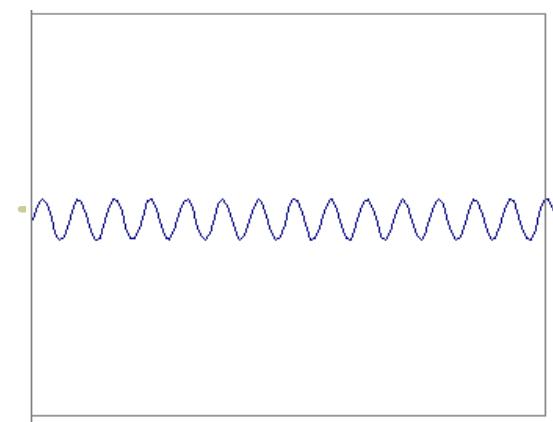
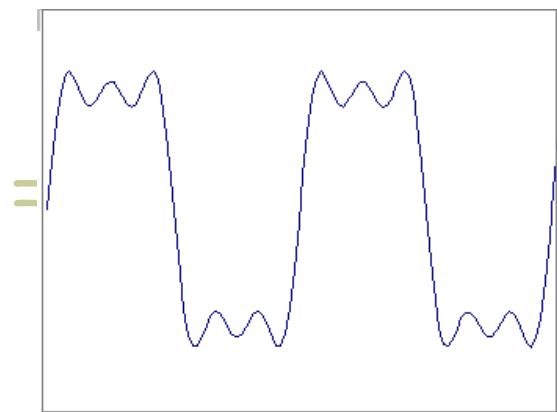
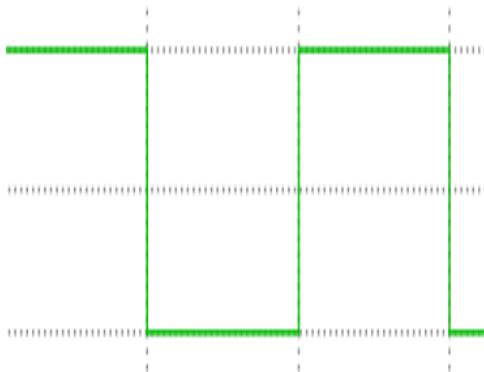


Frequency Spectra



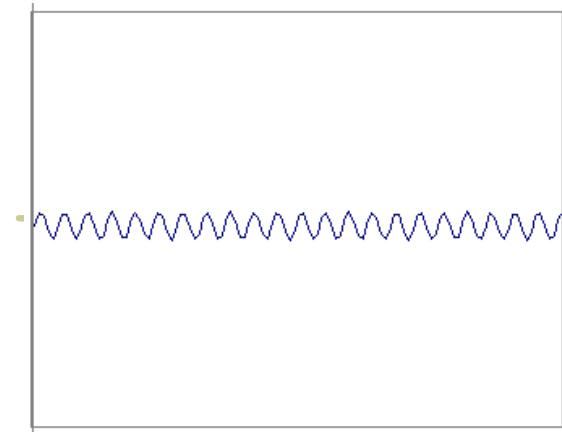
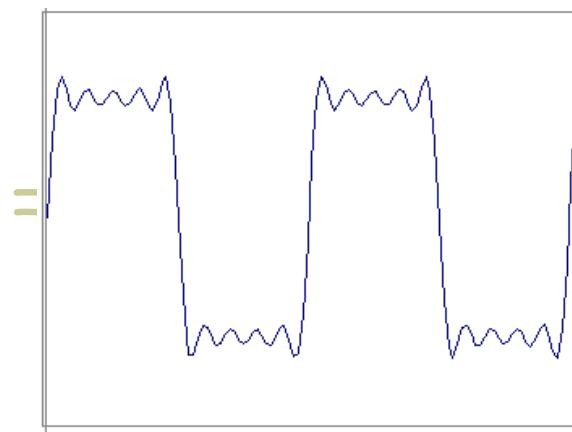
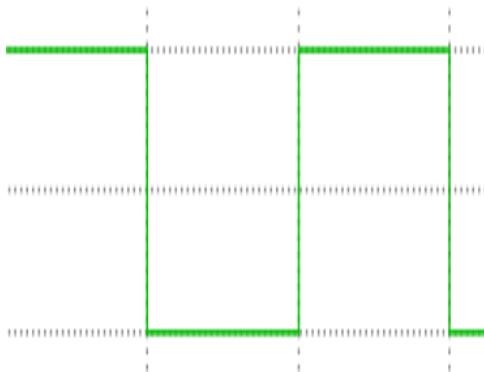


Frequency Spectra



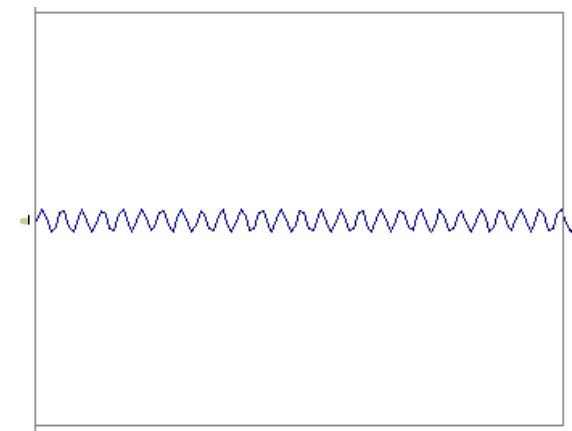
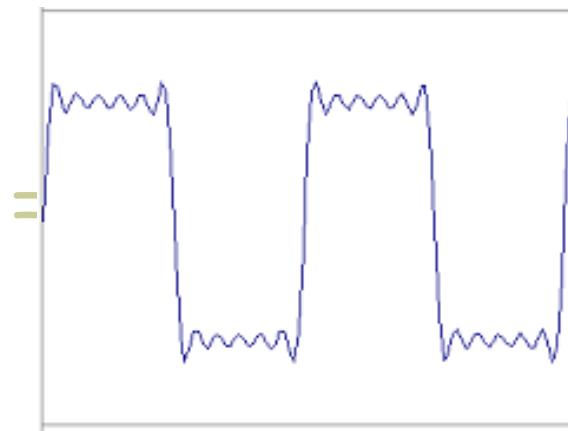
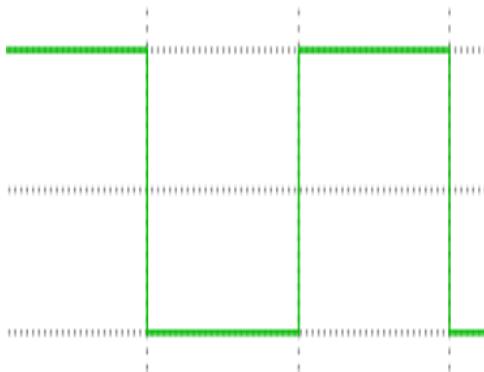


Frequency Spectra



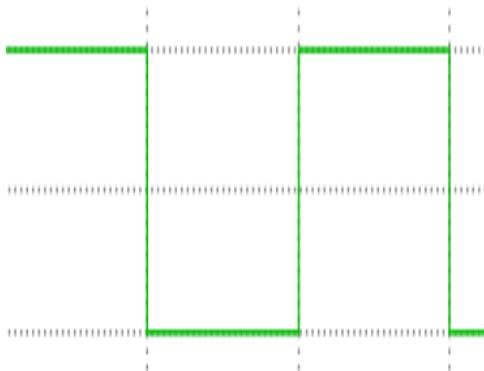


Frequency Spectra

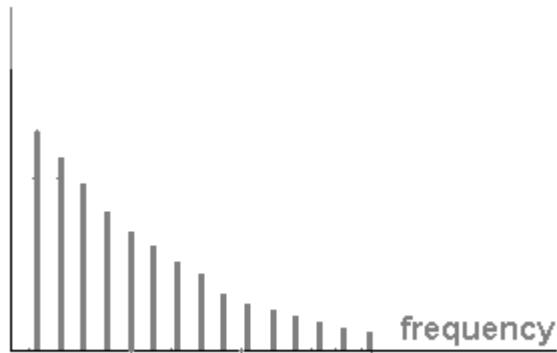




Frequency Spectra

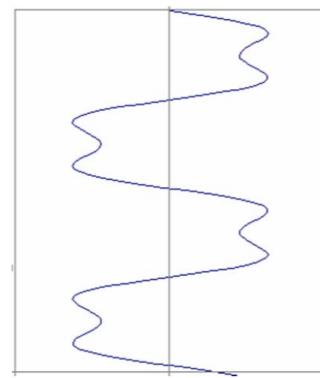
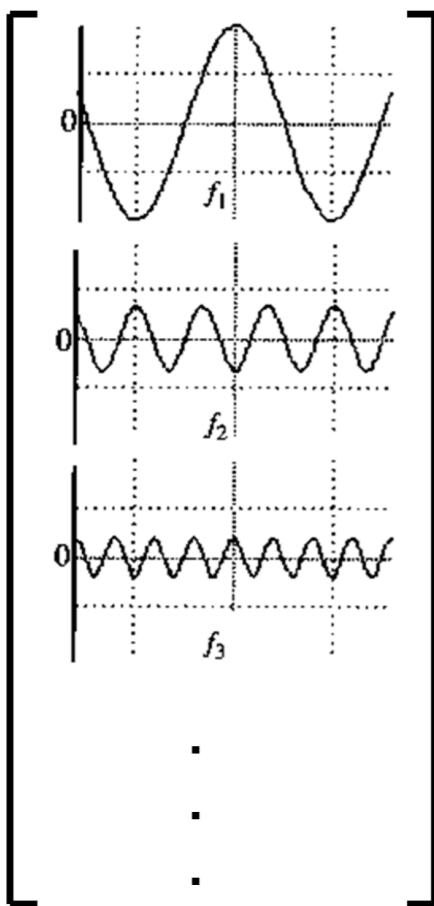


$$= A \sum_{k=1}^{\infty} \frac{1}{k} \sin(2\pi kt)$$

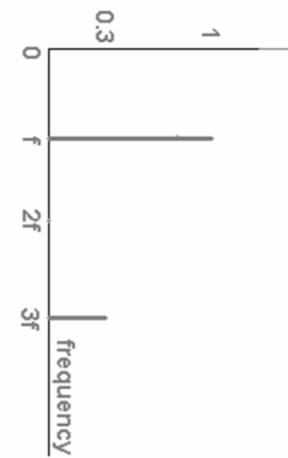


Fourier series: just a change of basis

$$\text{M } f(x) = F(\omega)$$

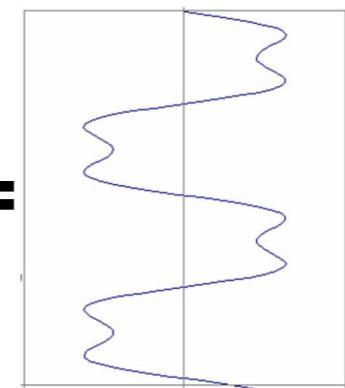
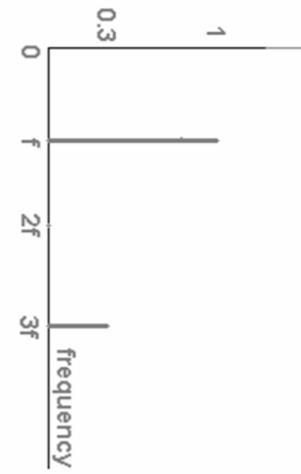
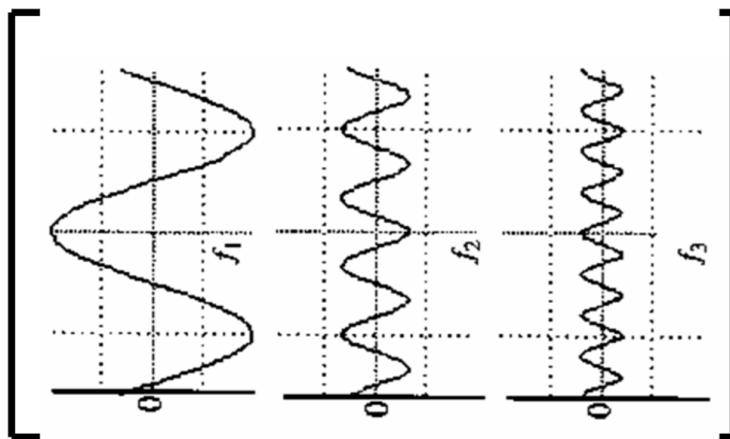


||



Inverse FT: Just a change of basis

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

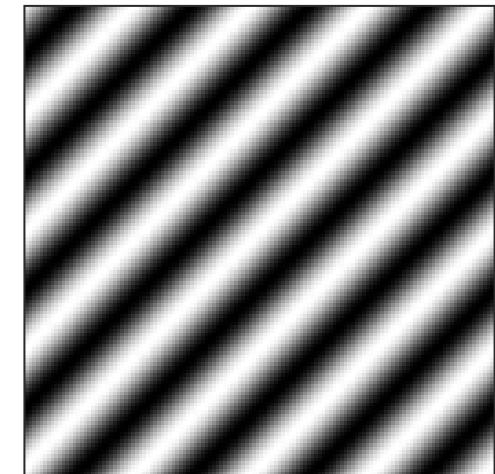
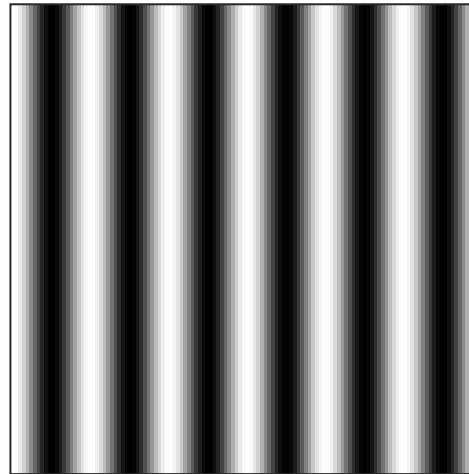
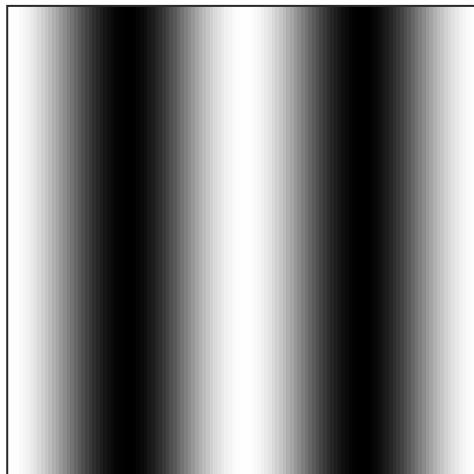




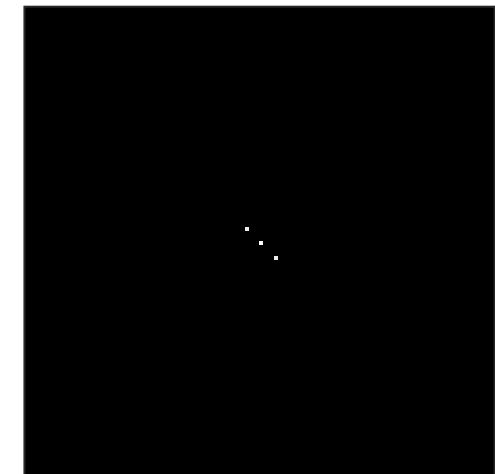
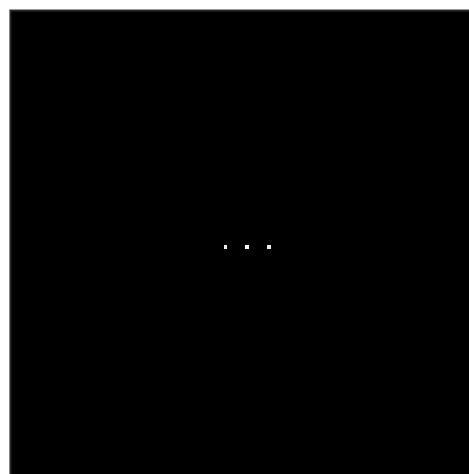
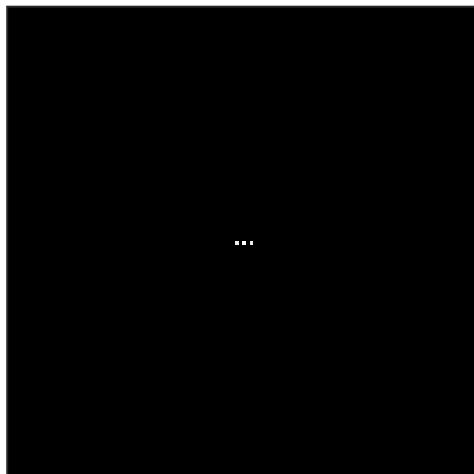
Fourier analysis in images



Intensity Image

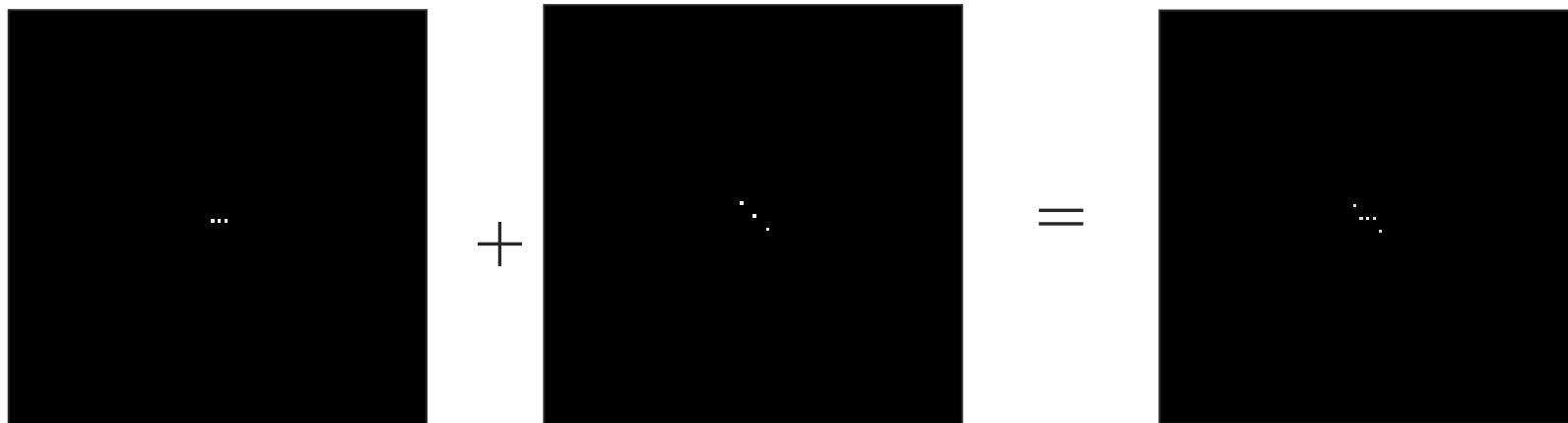
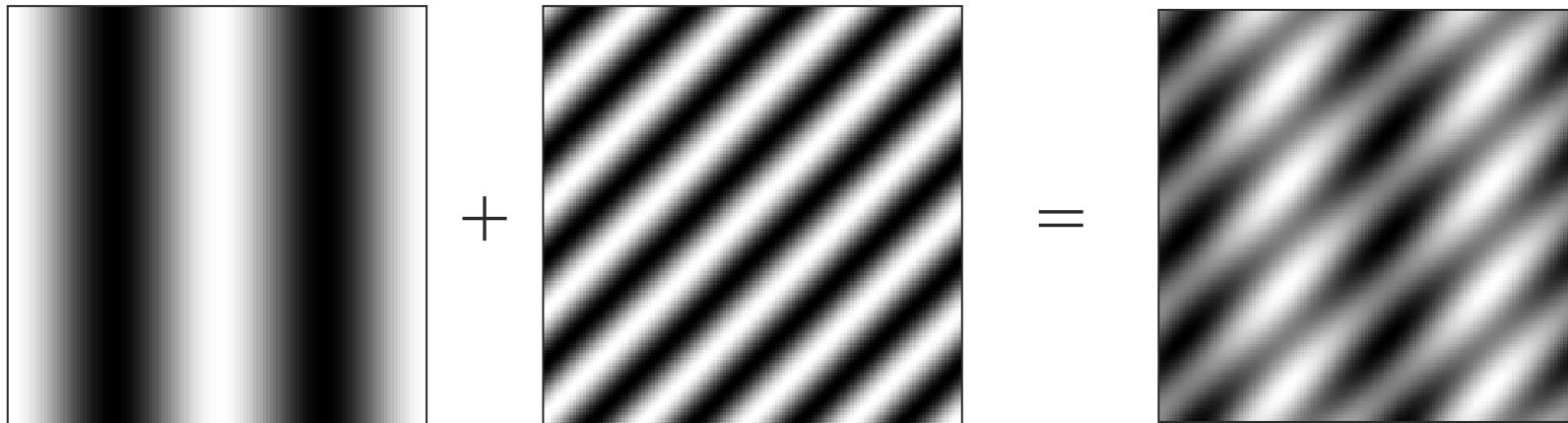


Fourier Image





Signals can be composed



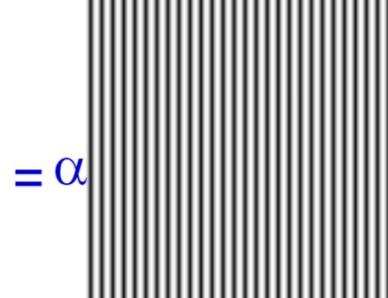
Summary

The spatial function $f(x, y)$

$$f(x, y) = \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} F(u, v) e^{j2\pi(ux+vy)} du dv$$

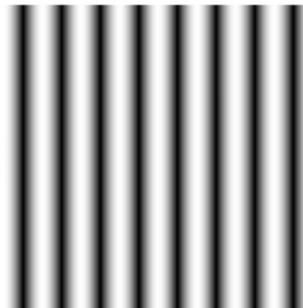
is decomposed into a weighted sum of 2D orthogonal basis functions in a similar manner to decomposing a vector onto a basis using scalar products.

$f(x, y)$

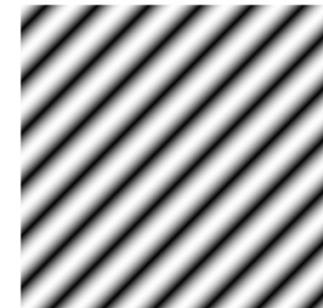


$= \alpha$

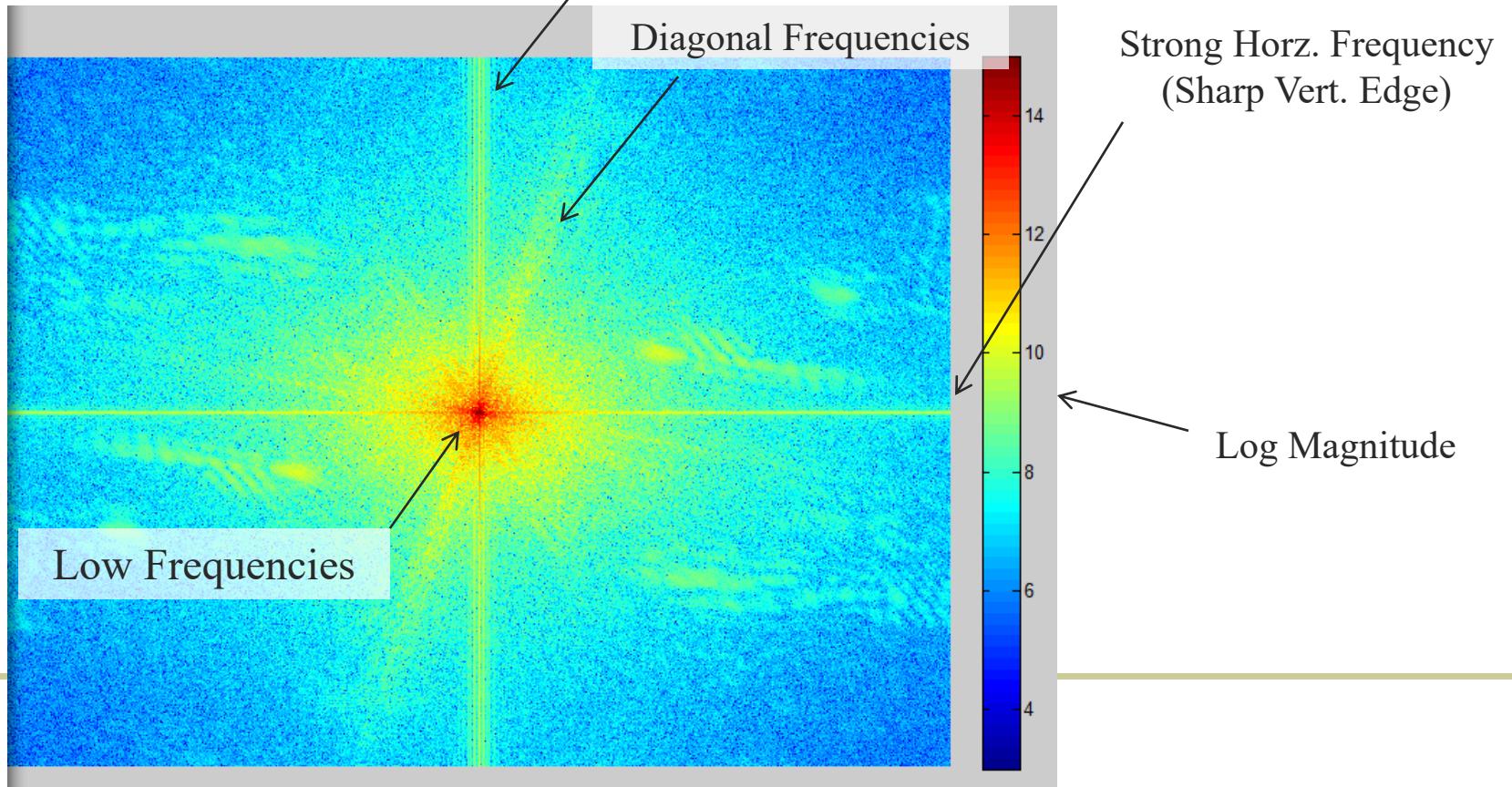
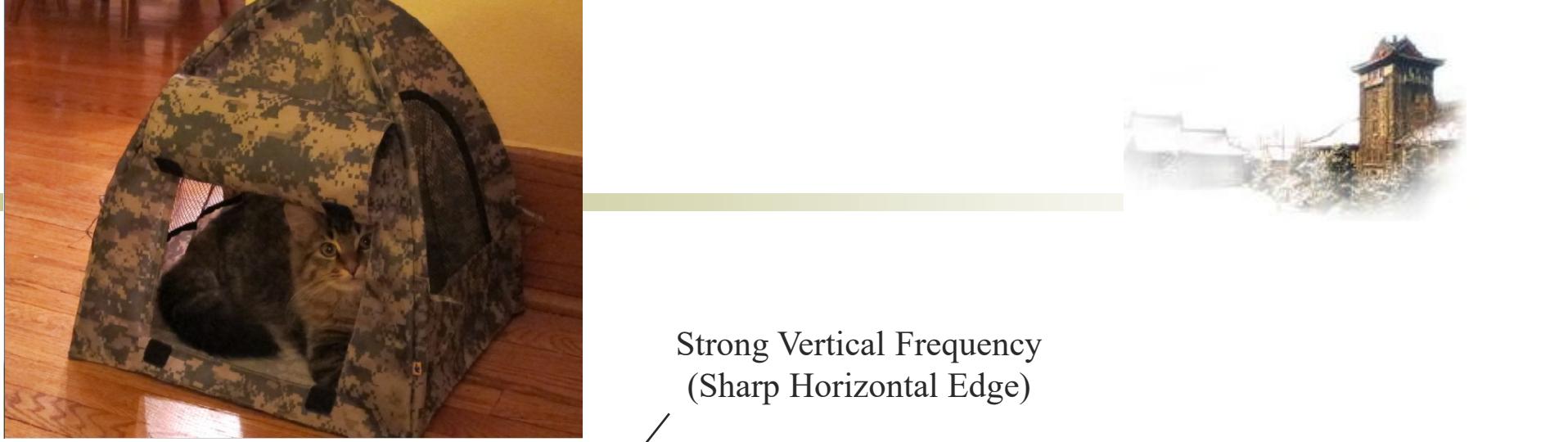
$+ \beta$



$+ \gamma$



$+ \dots$





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

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

Fast Fourier Transform (FFT): NlogN



Summary of Fourier Transform



$s(t)$ transforms (continuous-time)

	Continuous frequency	Discrete frequencies
Transform	$S(f) \triangleq \int_{-\infty}^{\infty} s(t) \cdot e^{-i2\pi ft} dt$	$\overbrace{\frac{1}{P} \cdot S\left(\frac{k}{P}\right)}^{S[k]} \triangleq \frac{1}{P} \int_{-\infty}^{\infty} s(t) \cdot e^{-i2\pi \frac{k}{P}t} dt \equiv \frac{1}{P} \int_P s_P(t) \cdot e^{-i2\pi \frac{k}{P}t} dt$
Inverse	$s(t) = \int_{-\infty}^{\infty} S(f) \cdot e^{i2\pi ft} df$	$s_P(t) = \underbrace{\sum_{k=-\infty}^{\infty} S[k] \cdot e^{i2\pi \frac{k}{P}t}}_{\text{Poisson summation formula (Fourier series)}}$

$s(nT)$ transforms (discrete-time)

	Continuous frequency	Discrete frequencies
Transform	$\underbrace{\frac{1}{T} S_{\frac{1}{T}}(f) \triangleq \sum_{n=-\infty}^{\infty} s(nT) \cdot e^{-i2\pi fnT}}_{\text{Poisson summation formula (DTFT)}}$	$\overbrace{\frac{1}{T} S_{\frac{1}{T}}\left(\frac{k}{NT}\right)}^{S[k]} \triangleq \sum_{n=-\infty}^{\infty} s(nT) \cdot e^{-i2\pi \frac{kn}{N}}$ $\equiv \underbrace{\sum_n s_P(nT) \cdot e^{-i2\pi \frac{kn}{N}}}_{\text{DFT}}$
Inverse	$s(nT) = T \int_{\frac{1}{T}} \frac{1}{T} S_{\frac{1}{T}}(f) \cdot e^{i2\pi fnT} df$ $\sum_{n=-\infty}^{\infty} s(nT) \cdot \delta(t - nT) = \underbrace{\int_{-\infty}^{\infty} \frac{1}{T} S_{\frac{1}{T}}(f) \cdot e^{i2\pi ft} df}_{\text{inverse Fourier transform}}$	$s_P(nT) = \overbrace{\frac{1}{N} \sum_k S[k] \cdot e^{i2\pi \frac{kn}{N}}}^{\text{inverse DFT}}$ $= \frac{1}{P} \sum_k S_{\frac{1}{T}}\left(\frac{k}{P}\right) \cdot e^{i2\pi \frac{kn}{N}}$

2-D DFT

变换对公式：1D→2D推广

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

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

频谱（幅度）

$$|F(u, v)| = \left[R^2(u, v) + I^2(u, v) \right]^{1/2}$$

相位角

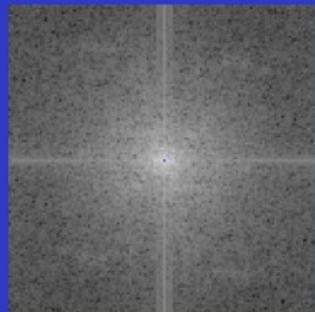
$$\phi(u, v) = \arctan[I(u, v)/R(u, v)]$$

功率谱

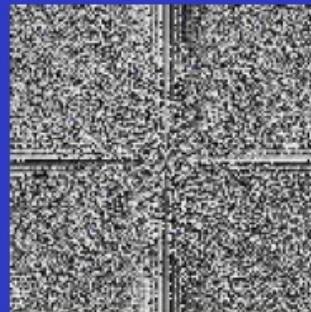
$$P(u, v) = |F(u, v)|^2 = R^2(u, v) + I^2(u, v)$$



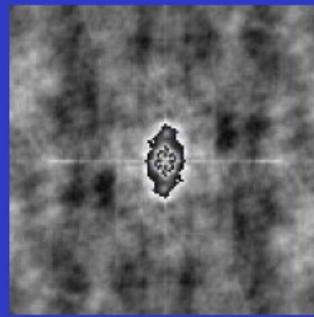
原图像



幅度谱



相位谱



由幅度谱重建

相位谱为0



由相位谱重建

幅度谱为常数



结论：相位谱可能具有更重要的应用

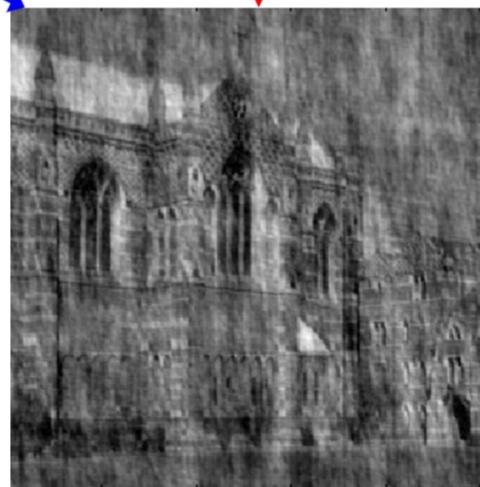
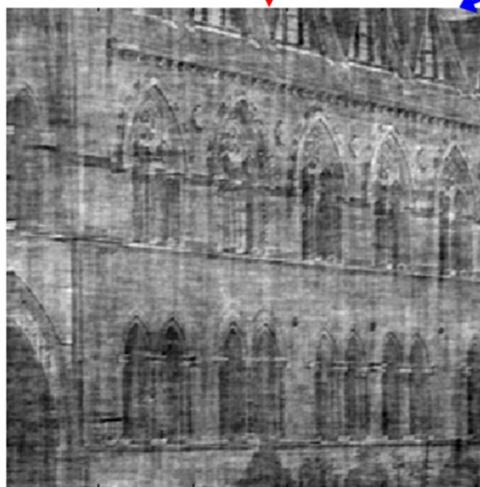
The importance of phase



phase

magnitude

phase



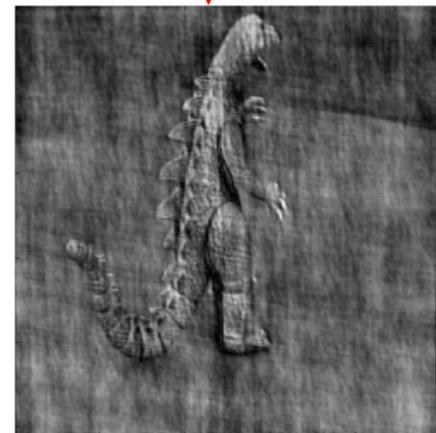
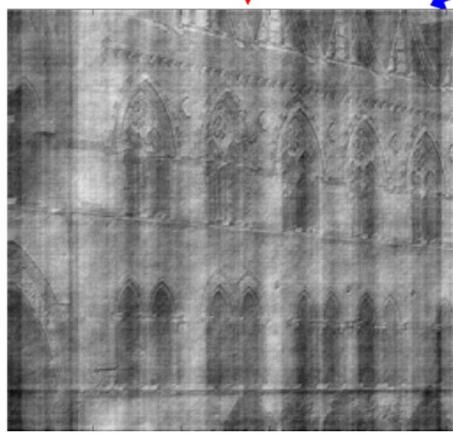
A second example



phase

magnitude

phase





The Convolution Theorem



- The Fourier transform of the convolution of two functions is the product of their Fourier transforms

$$\mathcal{F}[g * h] = \mathcal{F}[g]\mathcal{F}[h]$$

- The inverse Fourier transform of the product of two Fourier transforms is the convolution of the two inverse Fourier transforms

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

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



Properties of Fourier Transforms



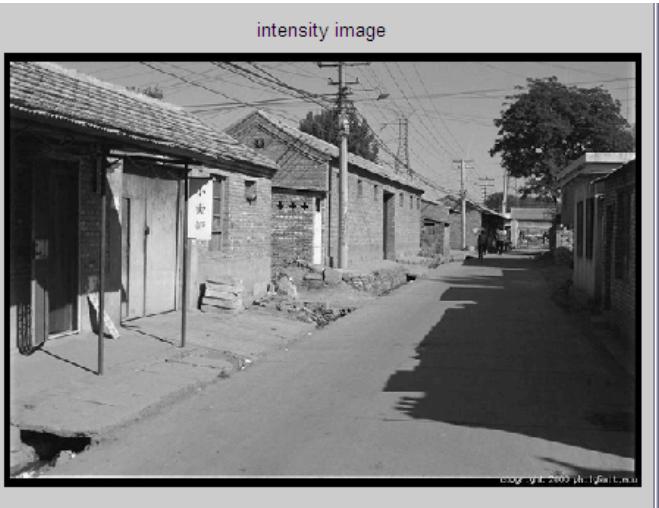
- Linearity $\mathcal{F}[ax(t) + by(t)] = a\mathcal{F}[x(t)] + b\mathcal{F}[y(t)]$
- Fourier transform of a real signal is symmetric about the origin
- The energy of the signal is the same as the energy of its Fourier transform



Filtering in spatial domain

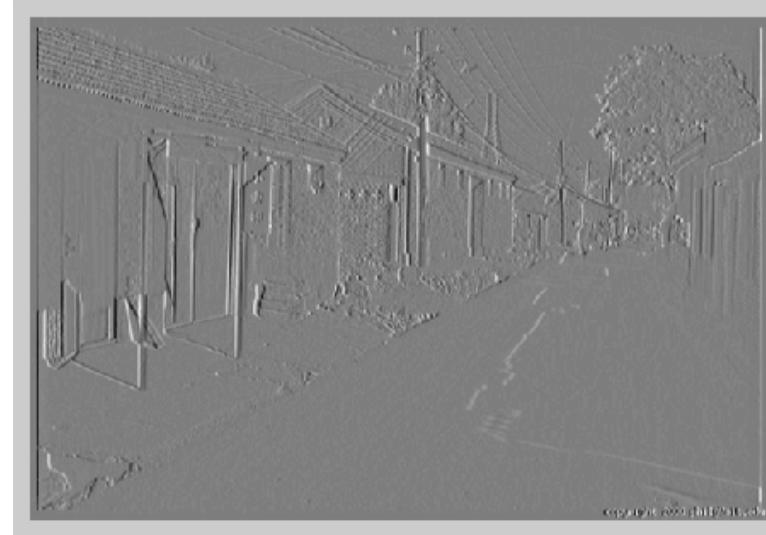


1	0	-1
2	0	-2
1	0	-1



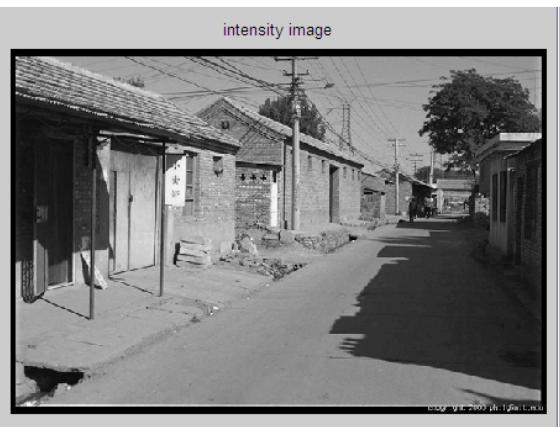
$$\ast \quad = \quad$$

A convolution operation diagram showing a 3x3 kernel (represented by a yellow asterisk) multiplied by a 3x3 input patch (represented by a gray square divided into nine smaller squares of decreasing gray levels), resulting in a single output value (represented by an equals sign).

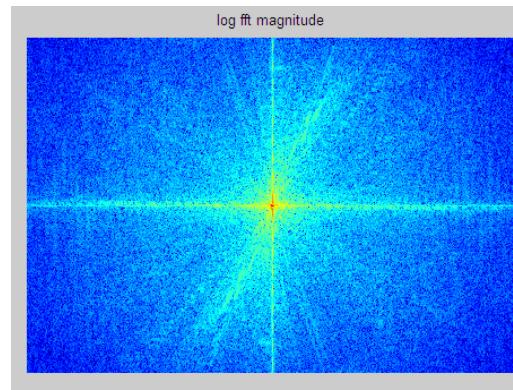




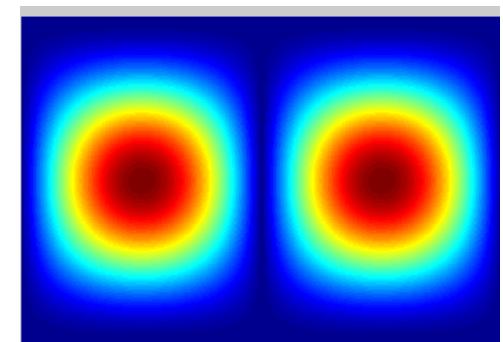
Filtering in frequency domain



FFT

A heatmap showing the log of the magnitude of the Fourier transform of the intensity image. The central point is bright yellow, and the surrounding frequencies decrease towards blue. A yellow arrow points from the intensity image to this stage.

X

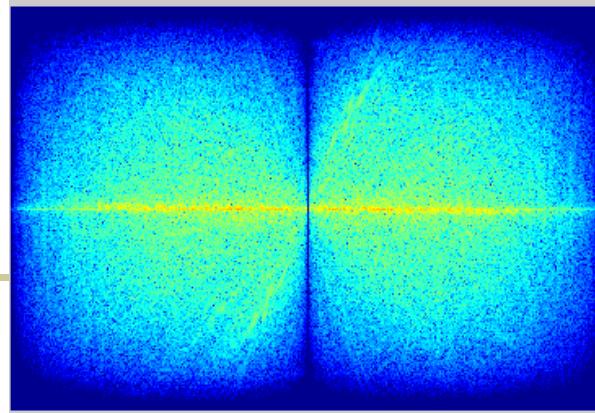


||

Inverse FFT

A heatmap showing the result of applying the filter in the frequency domain. The central peak is reduced, indicating the filtering effect. A yellow arrow points from this stage to the final output image.

A heatmap showing the result of applying the filter in the frequency domain. The central peak is reduced, indicating the filtering effect. A yellow arrow points from this stage to the final output image.



FFT

A heatmap showing the log of the magnitude of the Fourier transform of the intensity image. The central point is bright yellow, and the surrounding frequencies decrease towards blue. A yellow arrow points from the intensity image to this stage.

A heatmap showing the log of the magnitude of the Fourier transform of the intensity image. The central point is bright yellow, and the surrounding frequencies decrease towards blue. A yellow arrow points from the intensity image to this stage.





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)
```



Questions



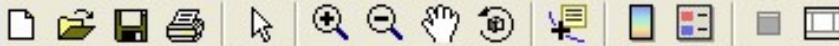
Which has more information, the phase or the magnitude?

What happens if you take the phase from one image and combine it with the magnitude from another image?

Figure 1



File Edit View Insert Tools Desktop Window Help



x 原图



y 原图



x 幅度谱 与 y 相位谱



y 幅度谱 与 x 相位谱



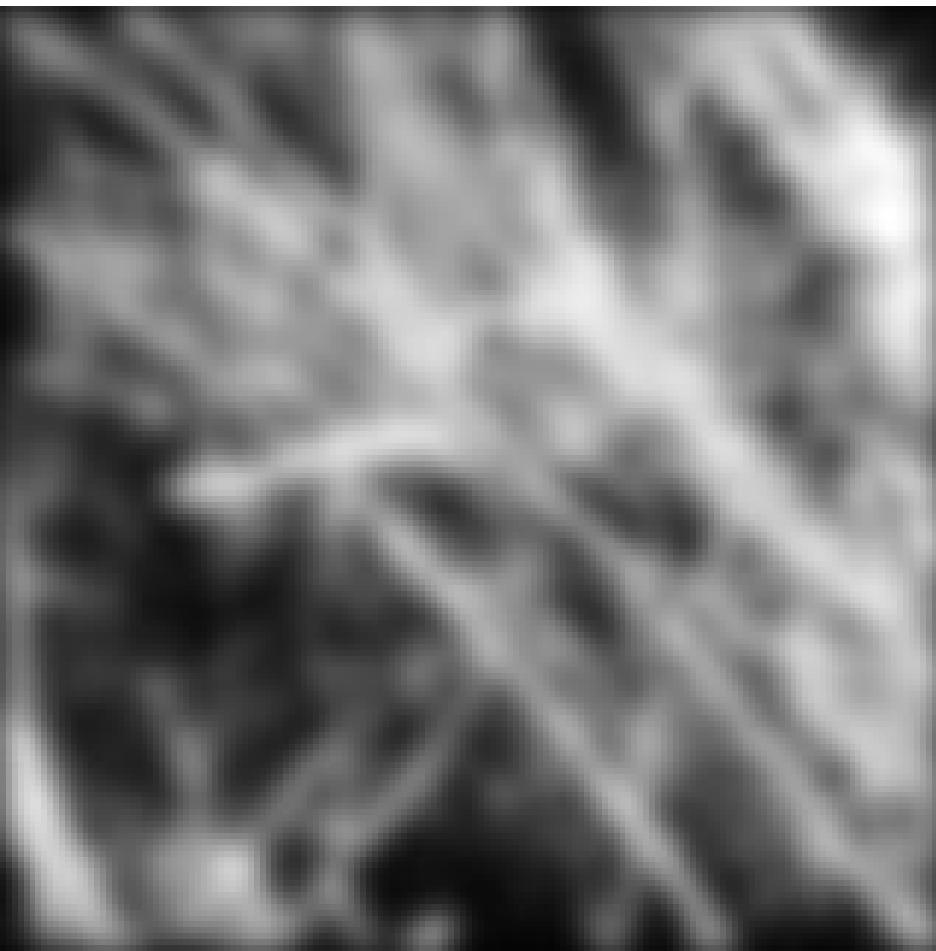


Filtering

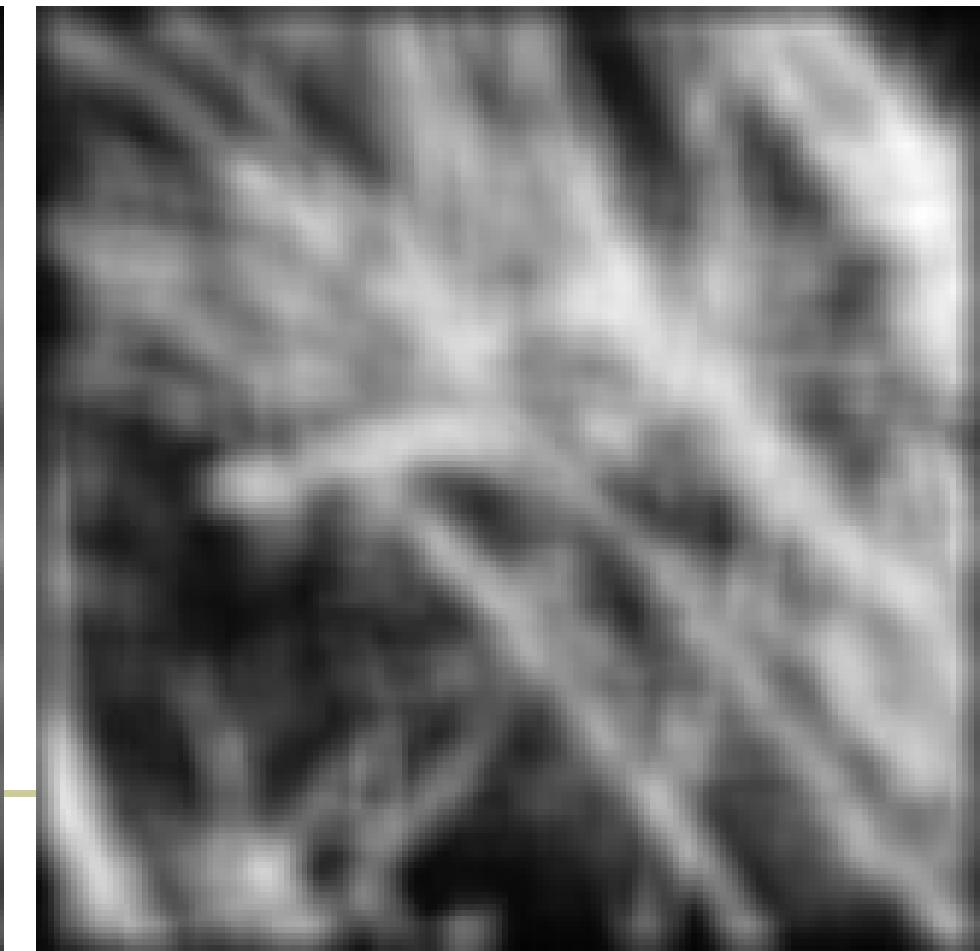


Why does the Gaussian give a nice smooth image, but the square filter give edgy artifacts?

Gaussian

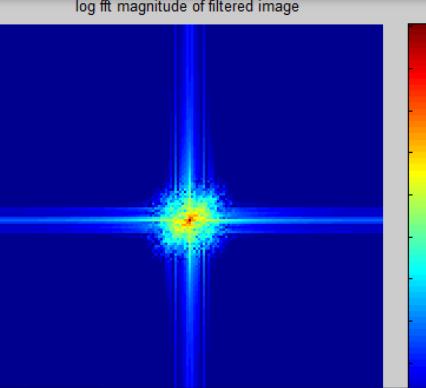
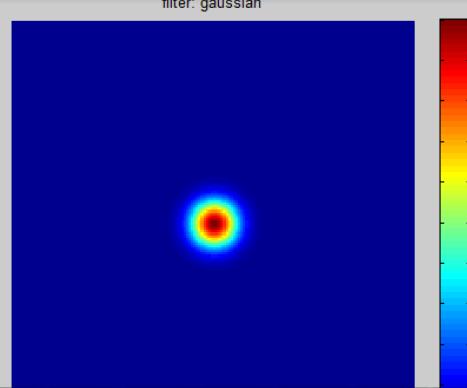
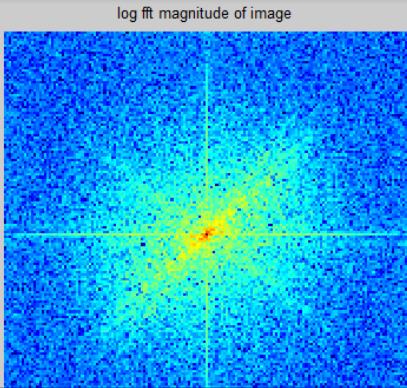
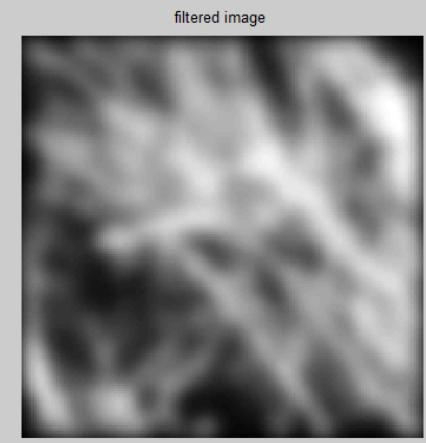
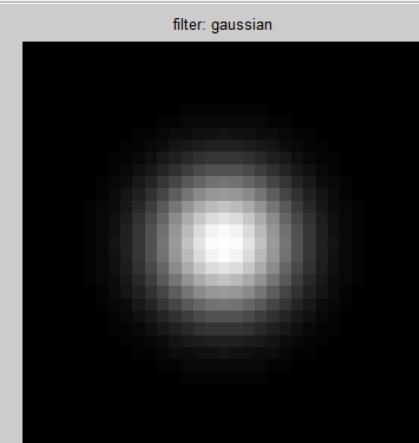
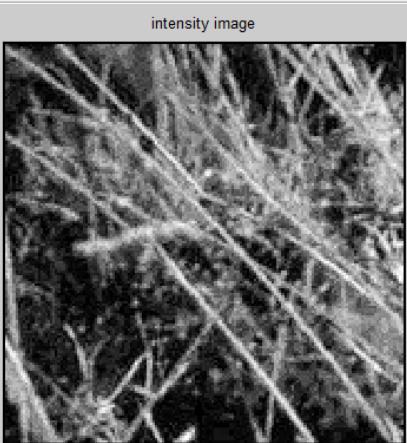


Box filter



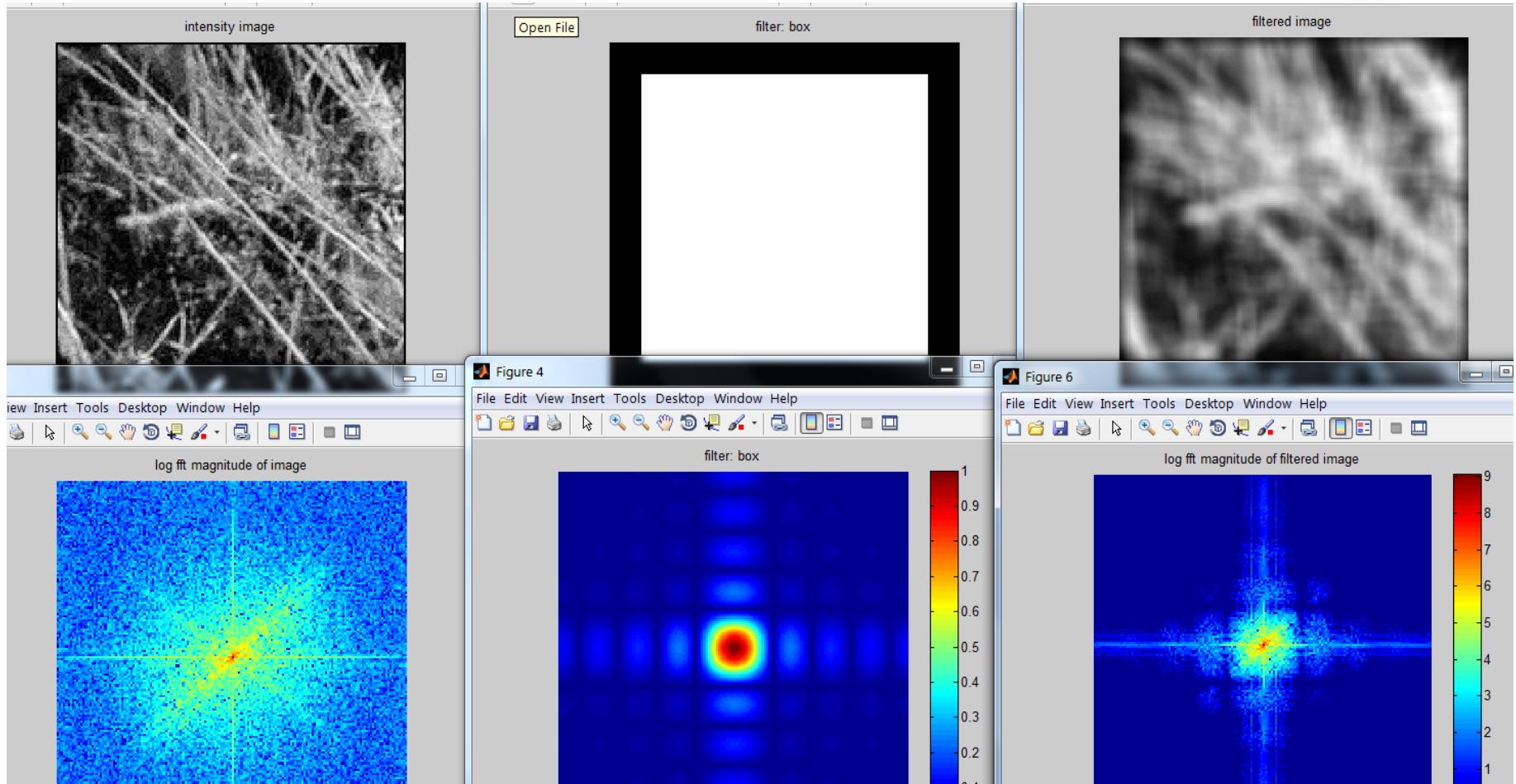


Gaussian Filter





Box Filter

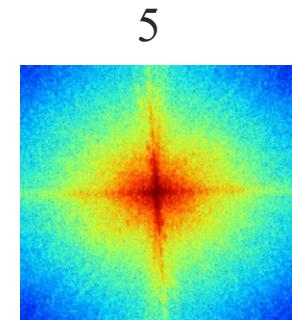
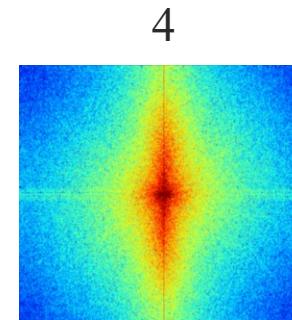
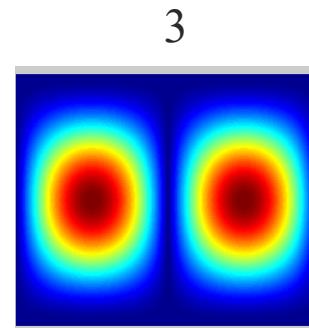
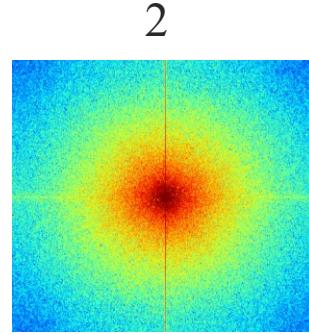
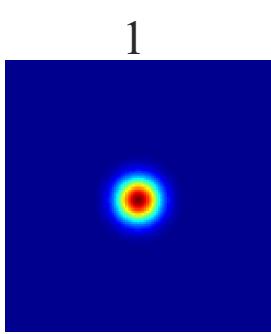




Question

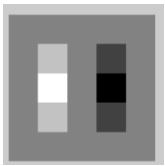


Match the spatial domain image to the Fourier magnitude image



B

A



D



E



Sampling

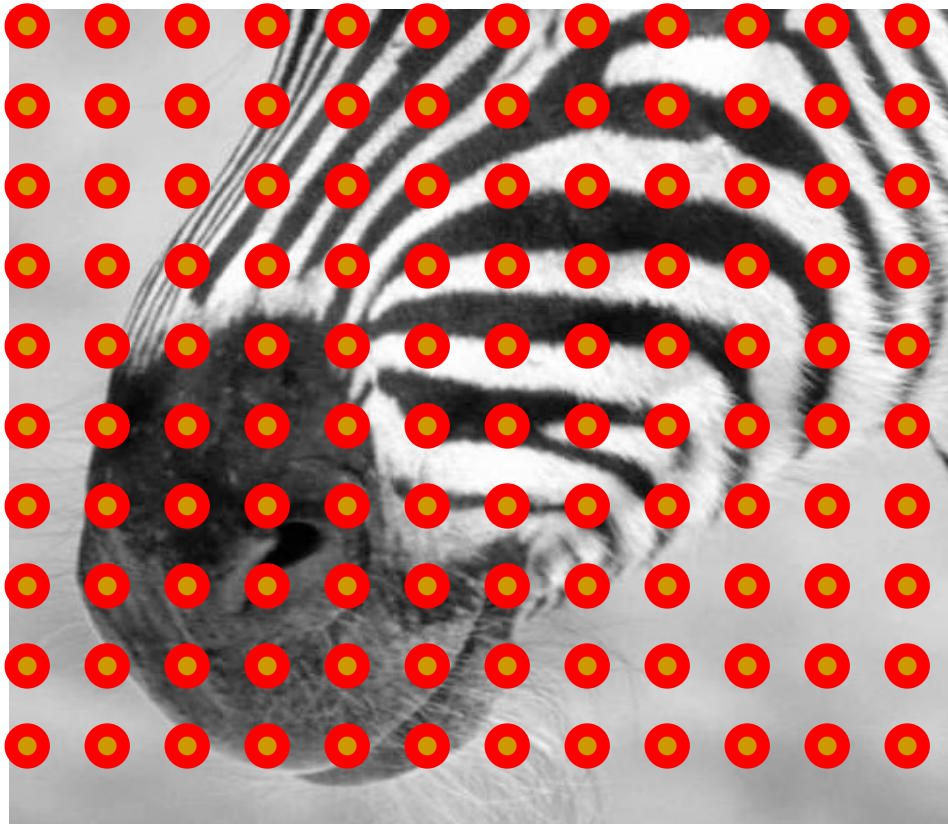


Why does a lower resolution image still make sense to us? What do we lose?





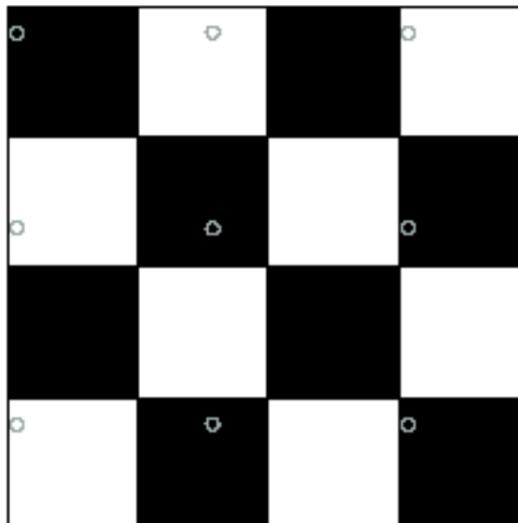
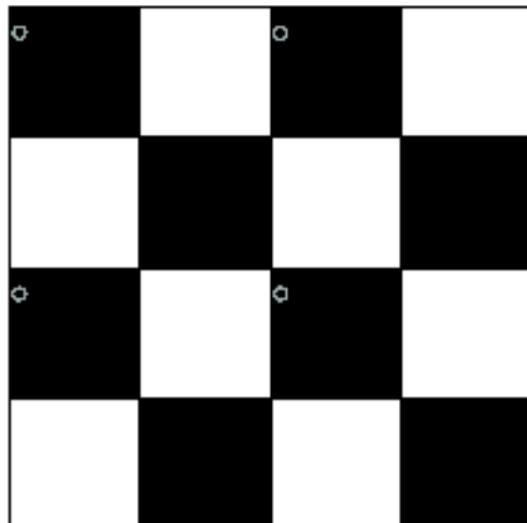
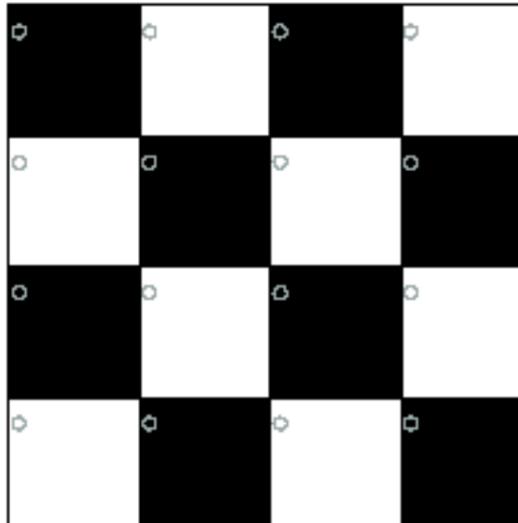
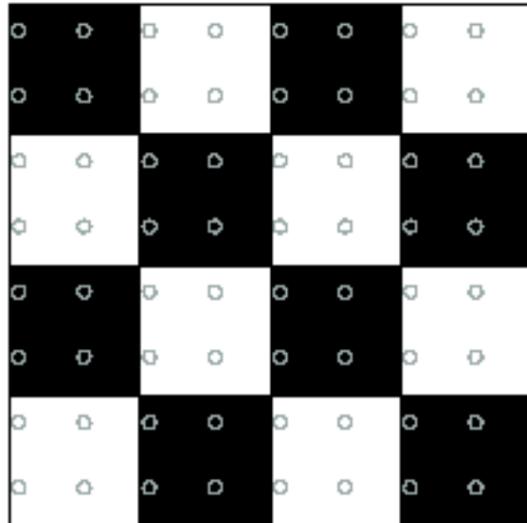
Subsampling by a factor of 2



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



How should we go about sampling



Let's resample the checkerboard by taking one sample at each circle.

In the top left board, the new representation is reasonable. Top right also yields a reasonable representation.

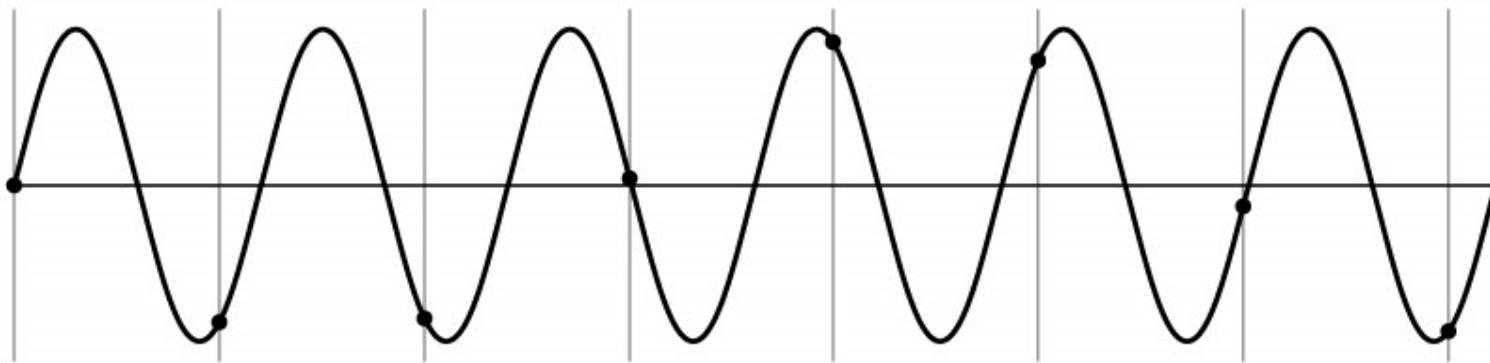
Bottom left is all black (dubious) and bottom right has checks that are too big.



How should we go about sampling



- 1D example (sinewave):

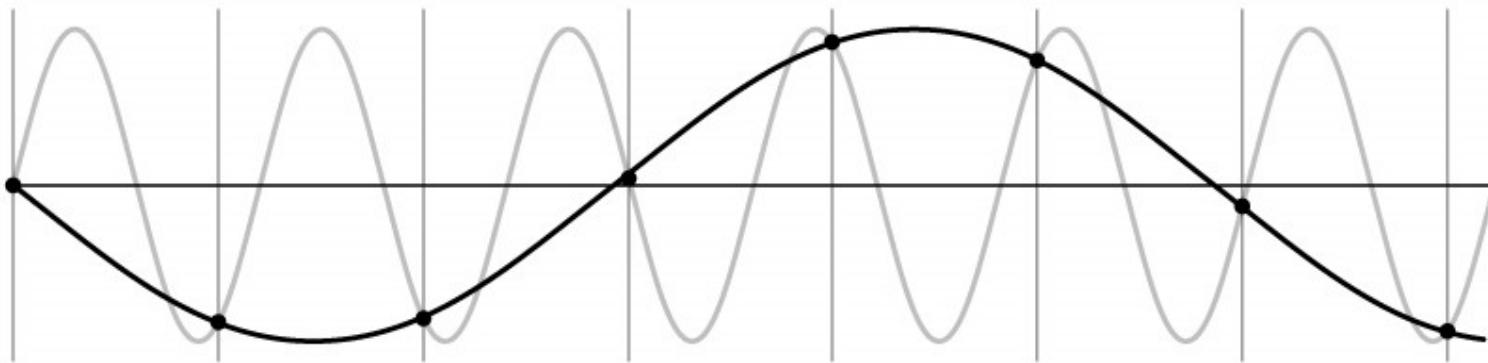




How should we go about sampling



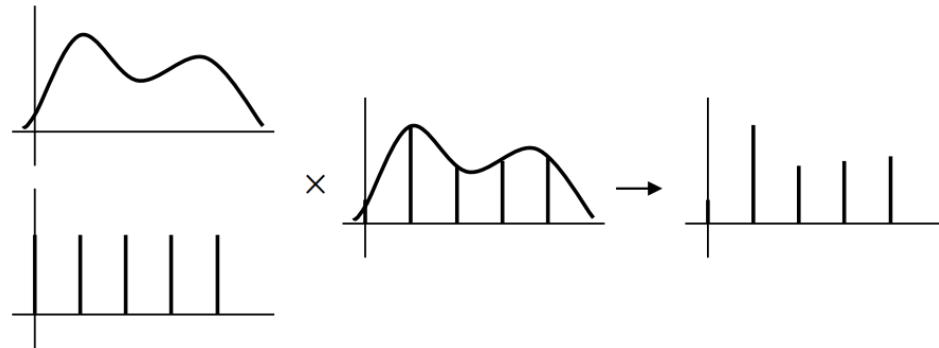
- 1D example (sinewave):





Fourier Interpretation: Sampling

- Sampling in the spatial domain is like multiplying with a spike function.



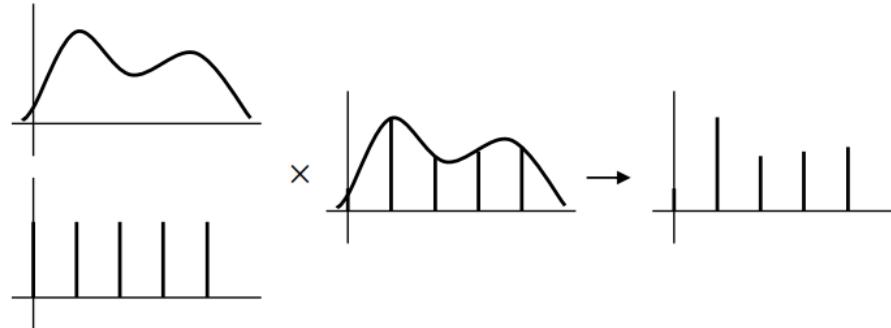
- Sampling in the frequency domain is like...

?

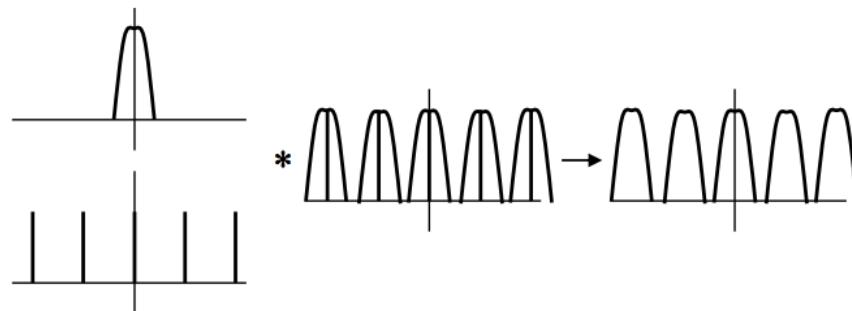


Fourier Interpretation: Sampling

- Sampling in the spatial domain is like multiplying with a spike function.

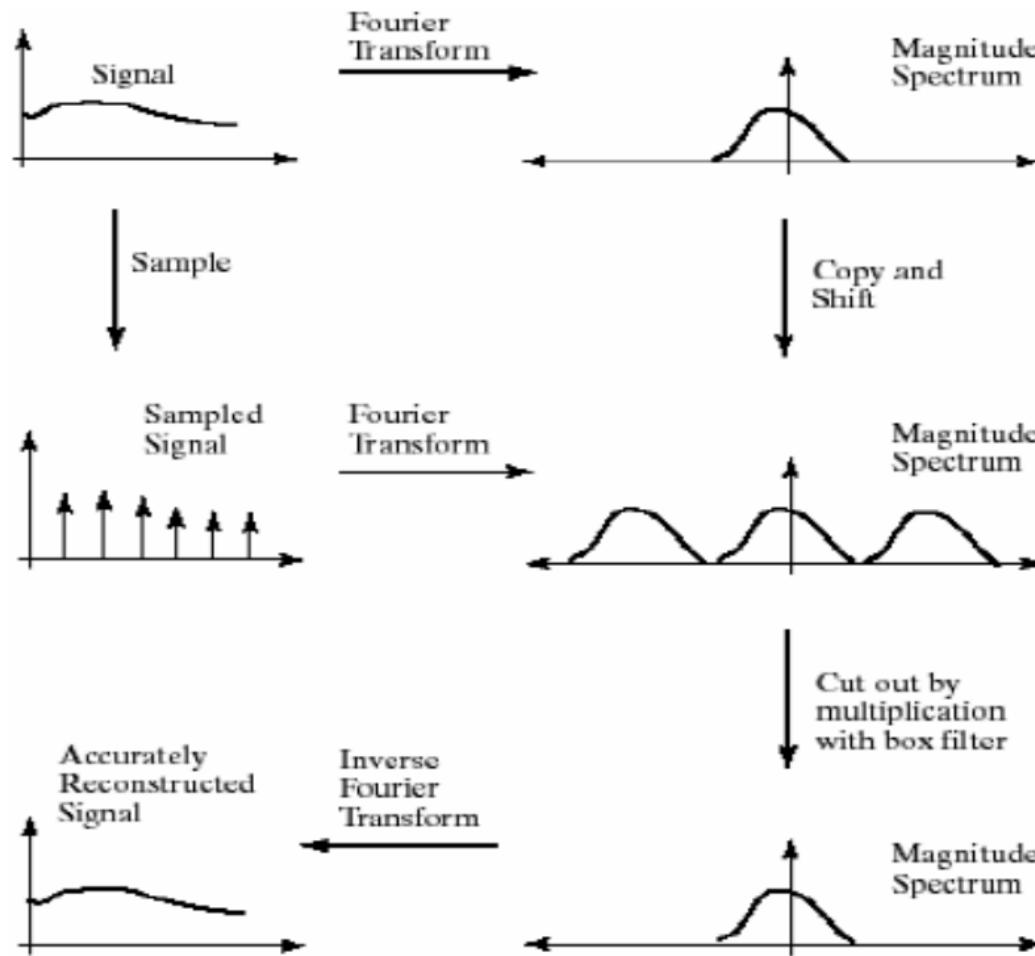


- Sampling in the frequency domain is like convolving with a spike function.



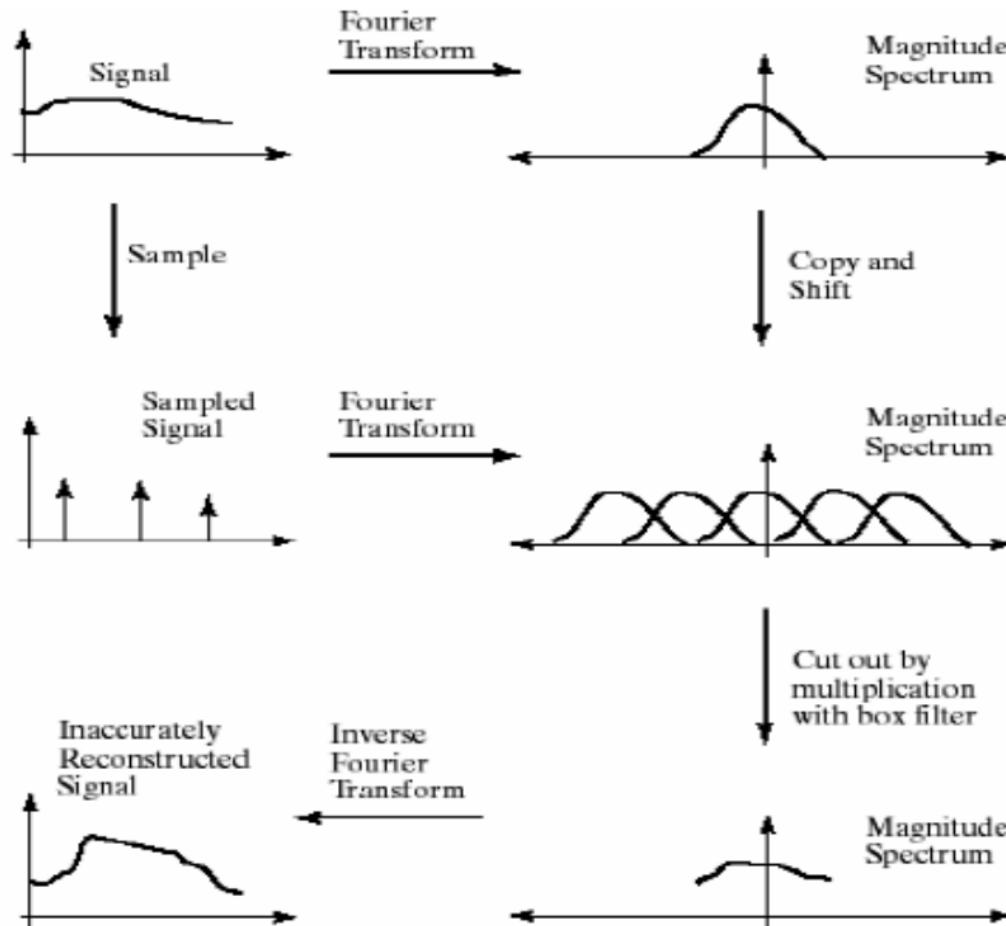


Fourier Interpretation: Sampling





Fourier Interpretation: Sampling





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”



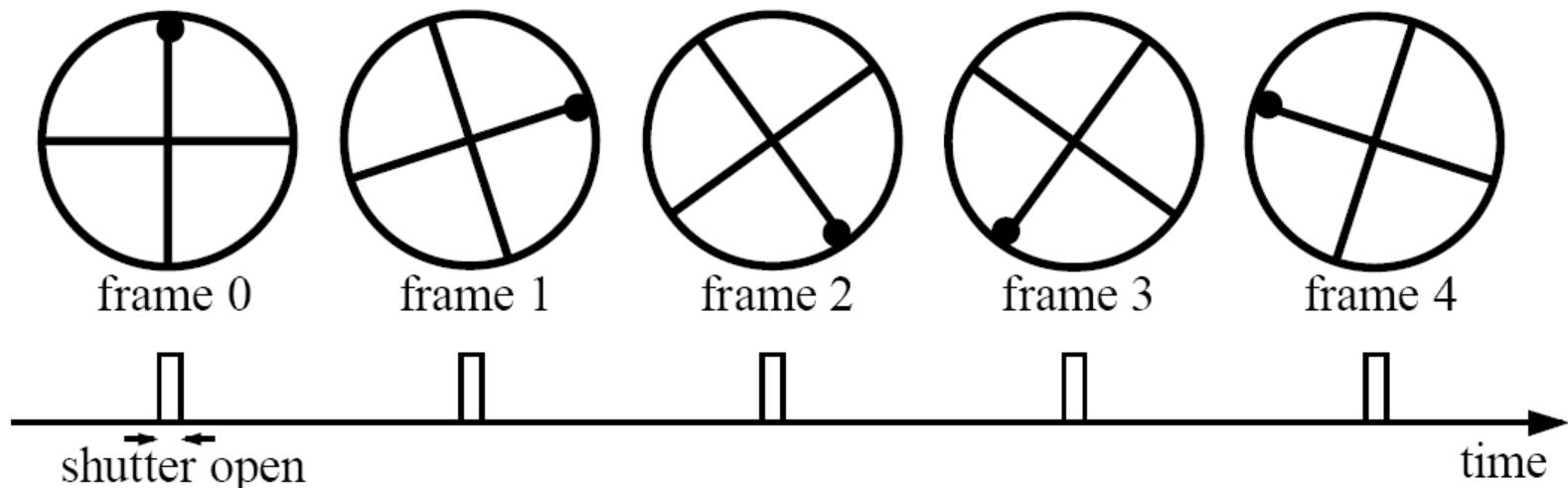
Aliasing in video



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)



Sampling and aliasing



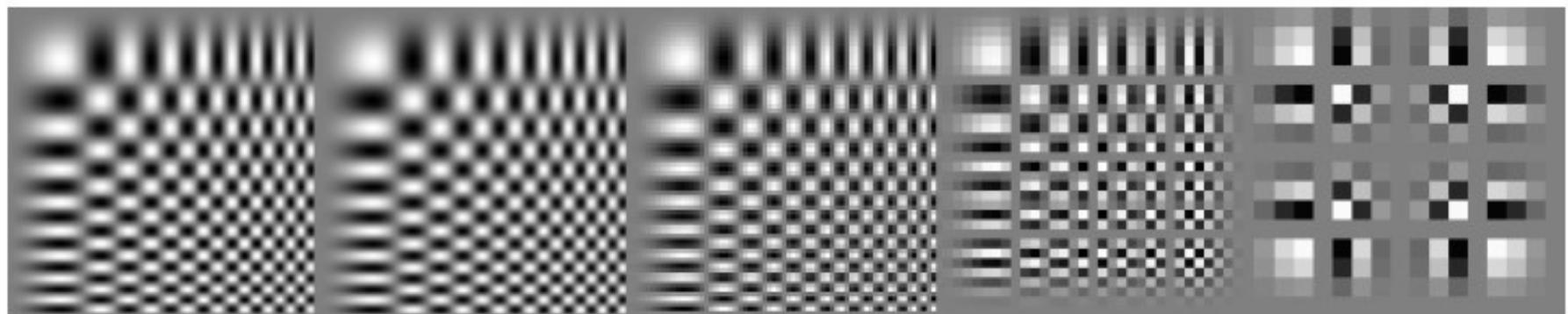
256x256

128x128

64x64

32x32

16x16

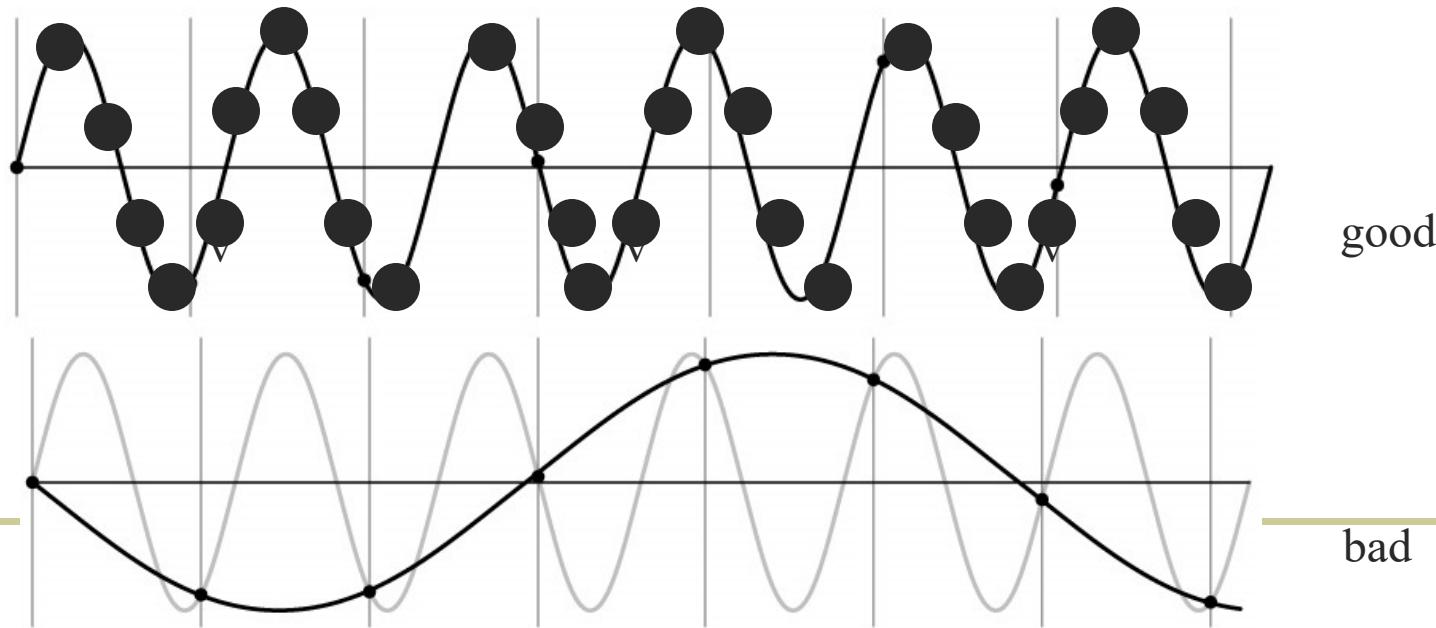




Nyquist-Shannon Sampling Theorem



- 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



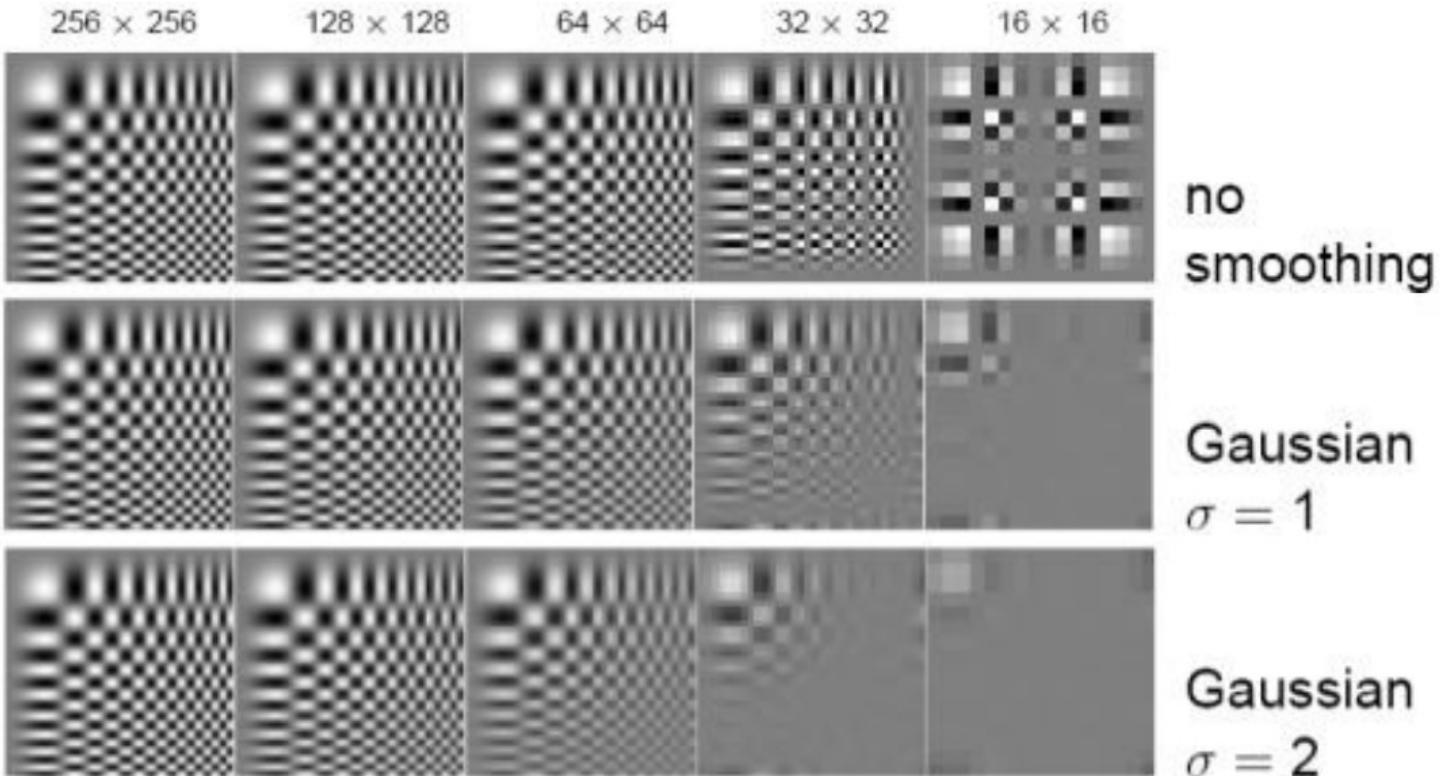
1. Start with $\text{image}(h, w)$
2. Apply low-pass filter

```
im.blur = imfilter(image, fspecial('gaussian', 7, 1))
```
3. Sample every other pixel

```
im.small = im.blur(1:2:end, 1:2:end);
```



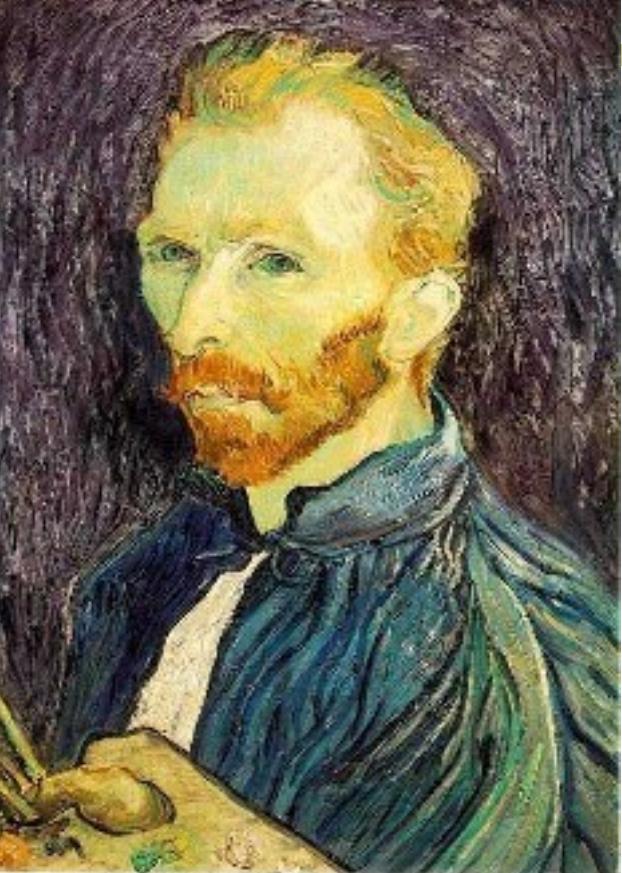
Anti-aliasing



Note: We cannot recover the high frequencies, but we can avoid artifacts by smoothing before resampling.



Subsampling without pre-filtering



1/2



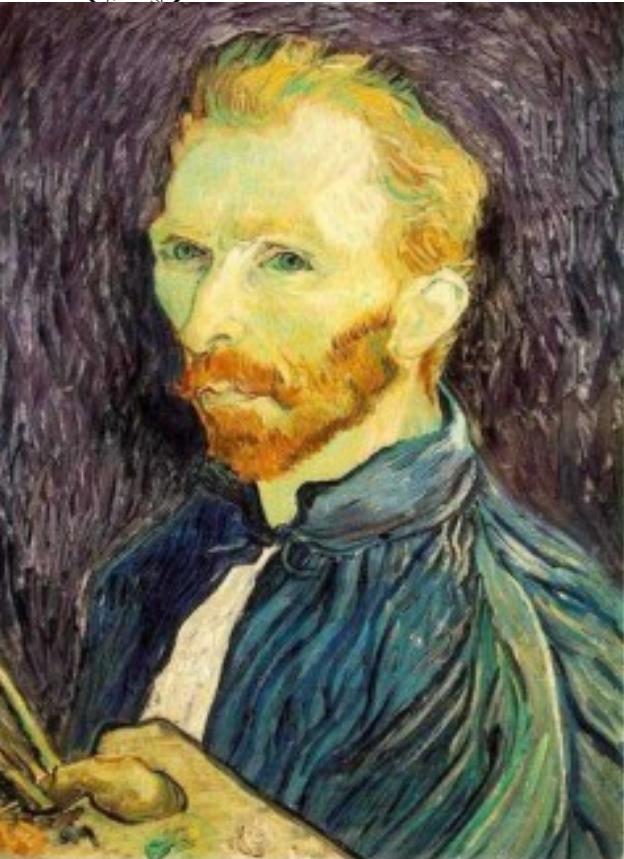
1/4 (2x zoom)



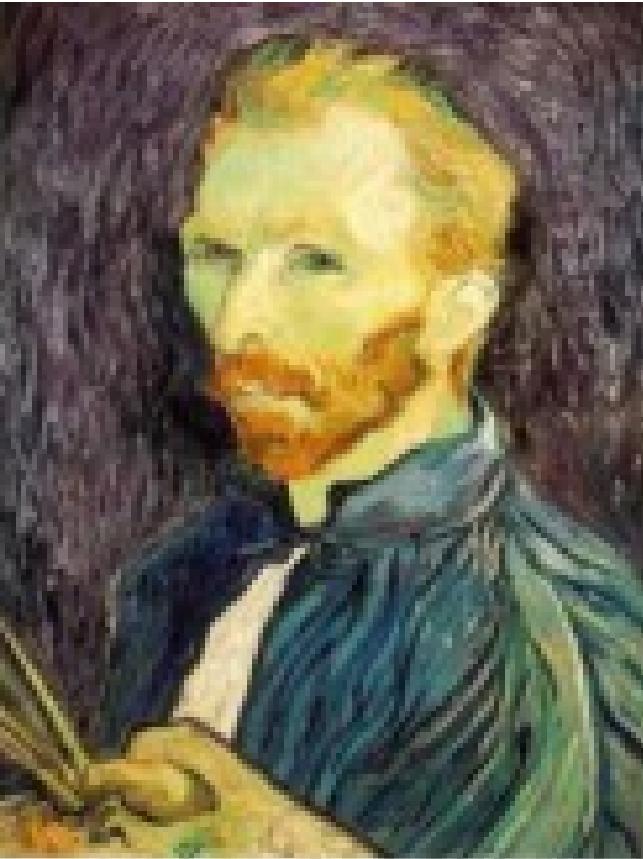
1/8 (4x zoom)



Subsampling with Gaussian pre-filtering



Gaussian 1/2



G 1/4



G 1/8

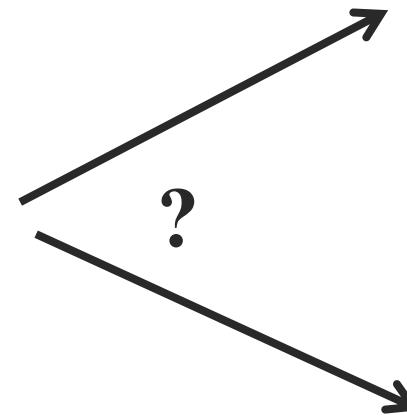


Why does a lower resolution image still make sense to us? What do we lose?





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

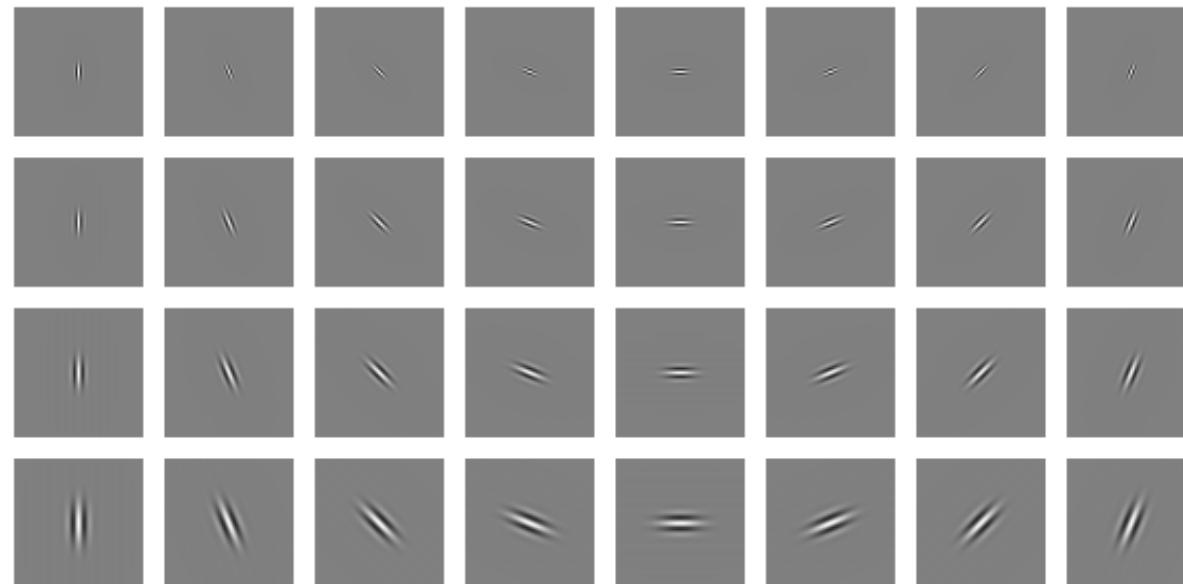




Clues from Human Perception

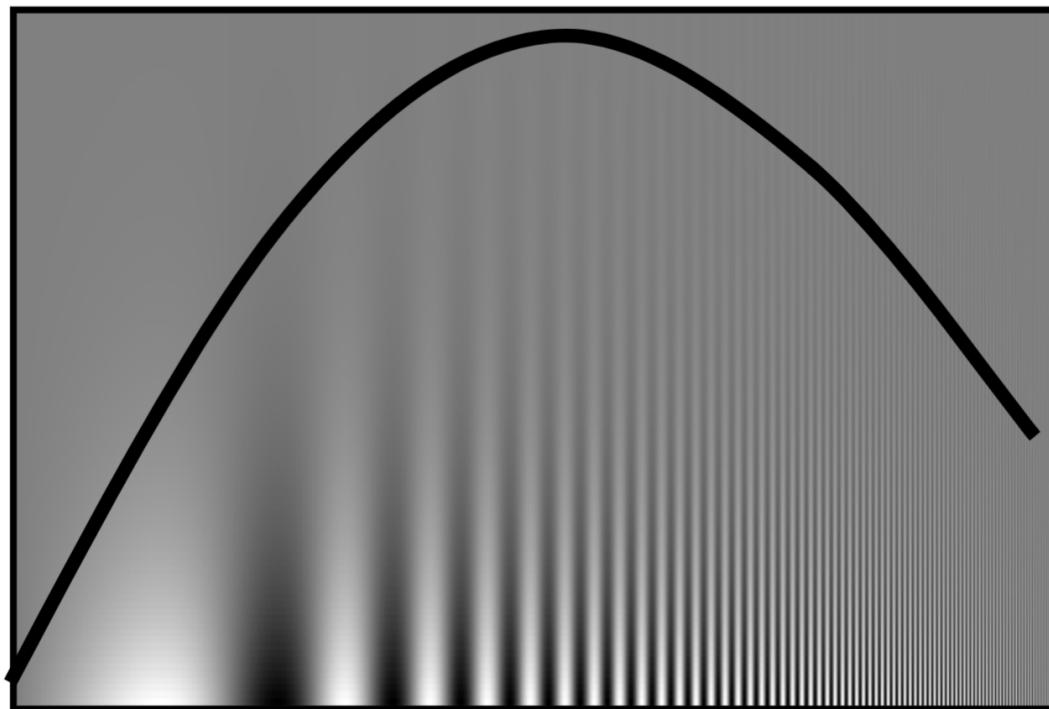


- Early processing in humans filters for various orientations and scales of frequency
- Perceptual cues in the mid-high frequencies dominate perception
- When we see an image from far away, we are effectively subsampling it



Early Visual Processing: Multi-scale edge and blob filters

Frequency Domain and Perception



Campbell-Robson contrast sensitivity curve

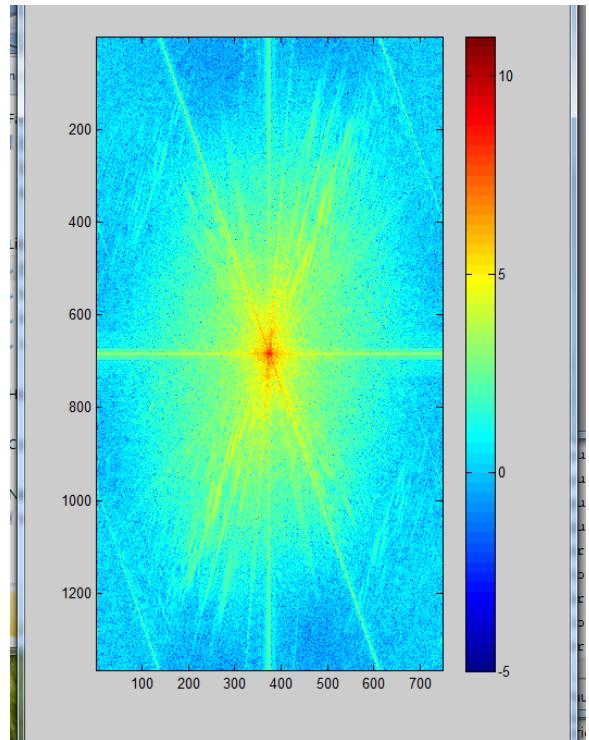
slide: A. Efros



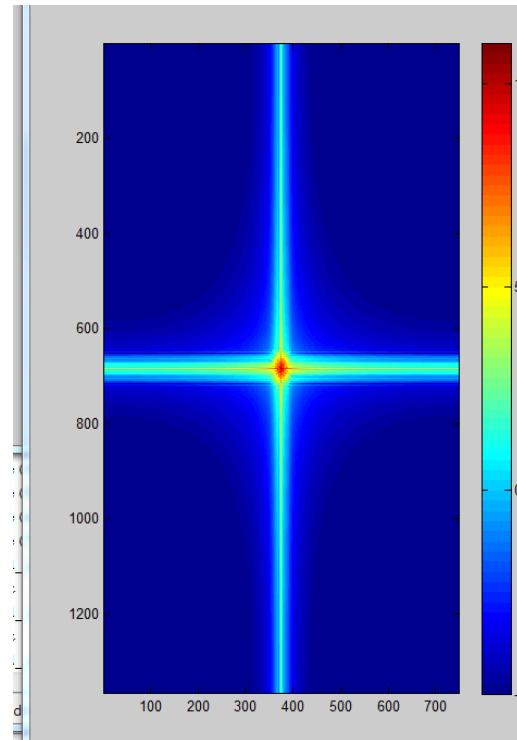
Hybrid Image in FFT



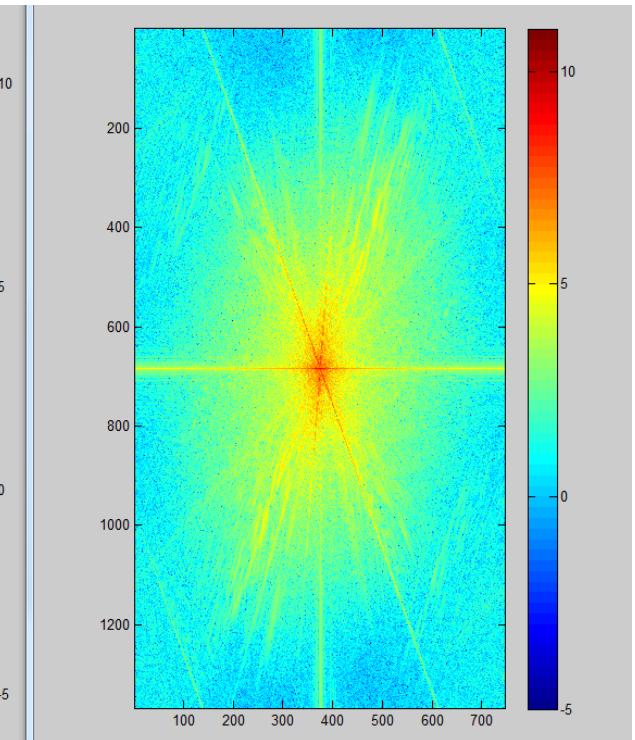
Hybrid Image



Low-passed Image



High-passed Image

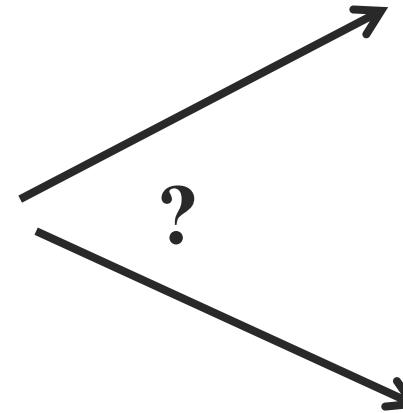




Perception



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





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)
- Remember to low-pass before sampling

