#### Reminder

- Project 1 due Wednesday
- Intended to be easy intro project
- What if your filtering is slow?
  - Separable filters (Gaussian)
  - Laplacian Pyramid
- What if my results aren't convincing?



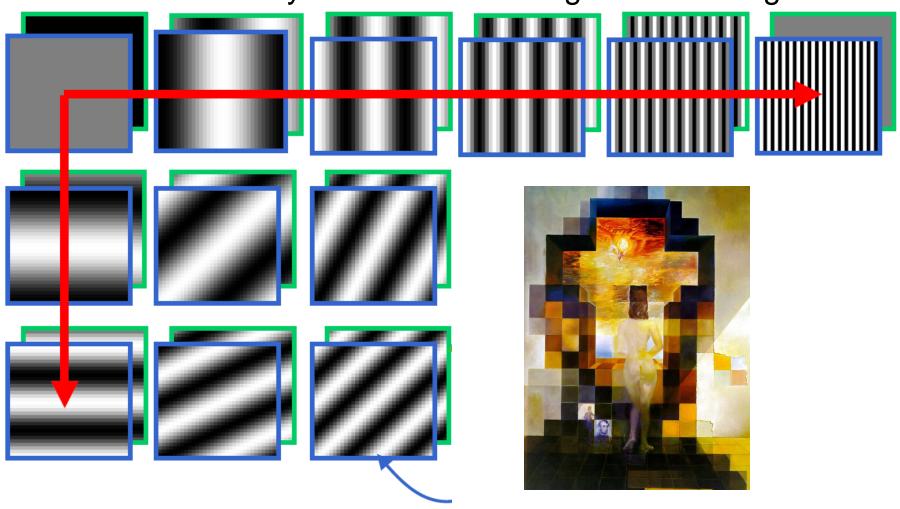
Recap: Fourier domain

#### 2d Fourier Transform

$$\hat{F}(k,\ell) = \frac{1}{\sqrt{MN}} \sum_{m=0}^{M-1} \sum_{n=0}^{N-1} F(m,n) e^{-j2\pi \left(k\frac{m}{M} + \ell\frac{n}{N}\right)}$$

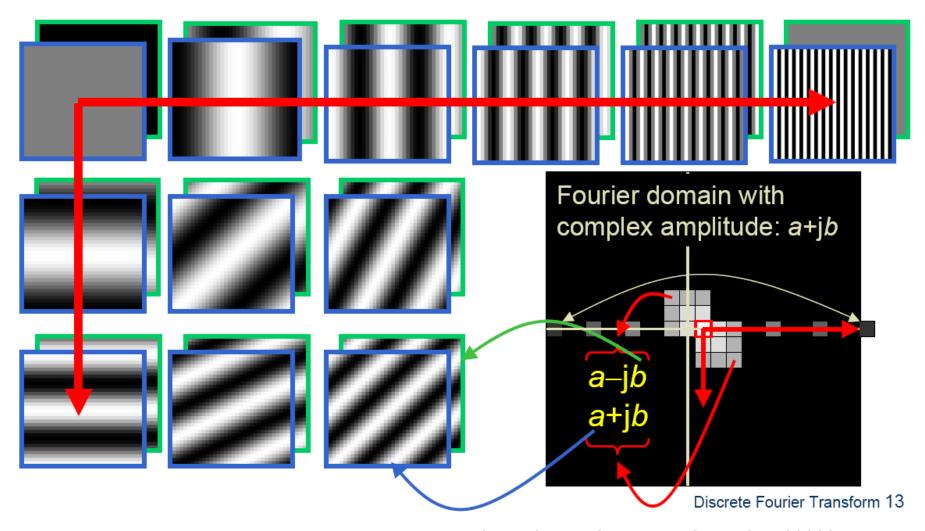
#### **Fourier Bases**

Teases away fast vs. slow changes in the image.



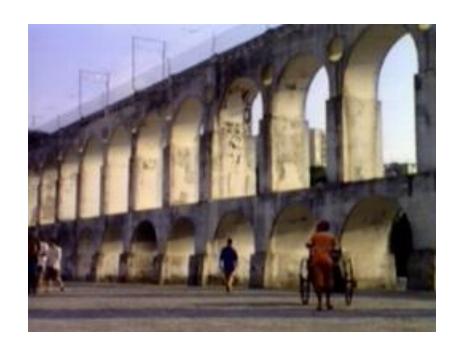
This change of basis is the Fourier Transform

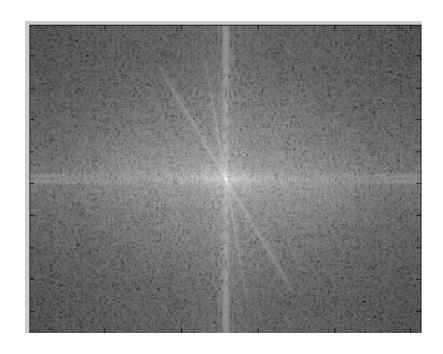
#### **Fourier Bases**



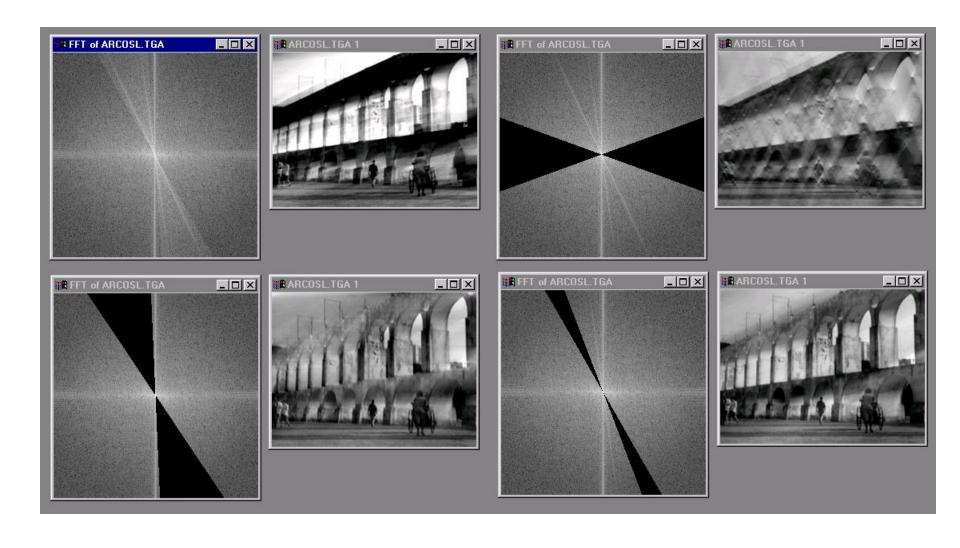
in Matlab, check out: imagesc(log(abs(fftshift(fft2(im)))));

## Man-made Scene

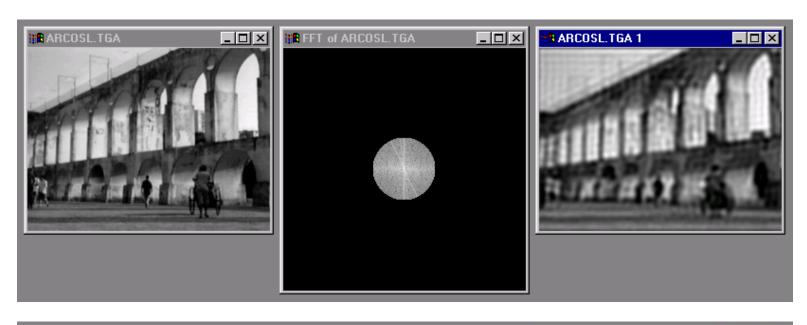


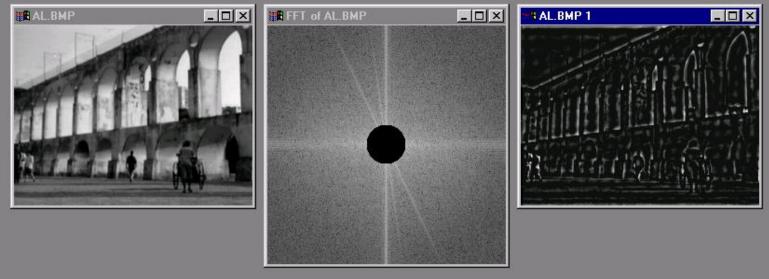


#### Can change spectrum, then reconstruct



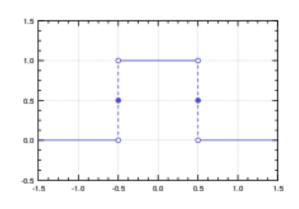
## Low and High Pass filtering



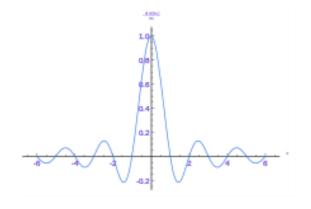


#### Sinc Filter

 What is the spatial representation of the hard cutoff in the frequency domain?



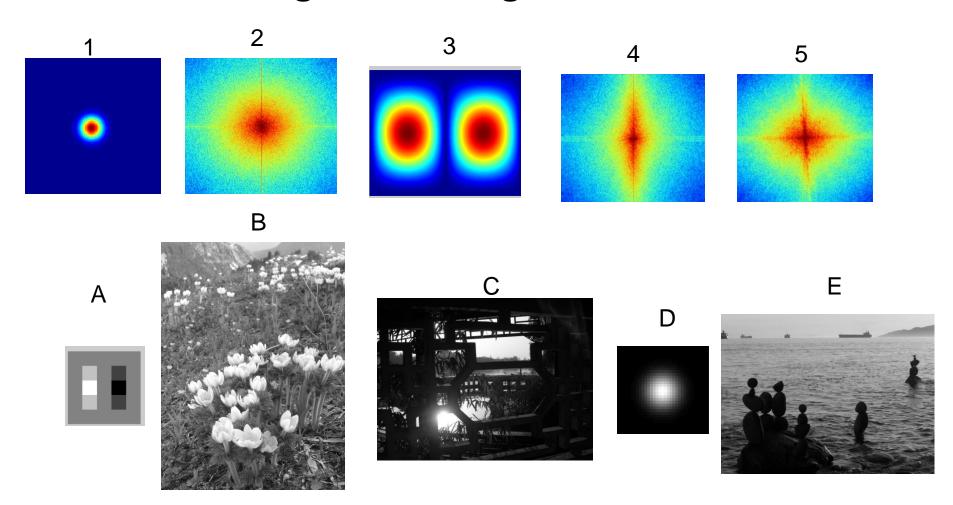
Frequency Domain



**Spatial Domain** 

#### Review

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



#### The Convolution Theorem

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

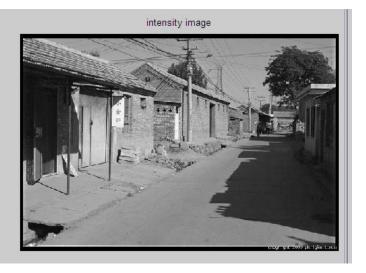
$$F[g * h] = F[g]F[h]$$

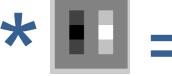
 Convolution in spatial domain is equivalent to multiplication in frequency domain!

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

## Filtering in spatial domain

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





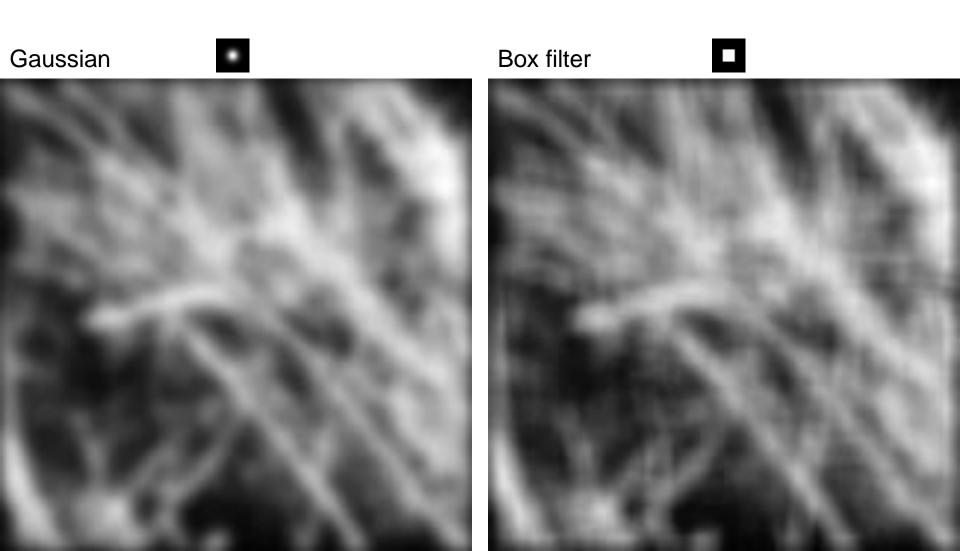


# Filtering in frequency domain **FFT** log fft magnitude FFT Inverse FFT Slide: Hoiem

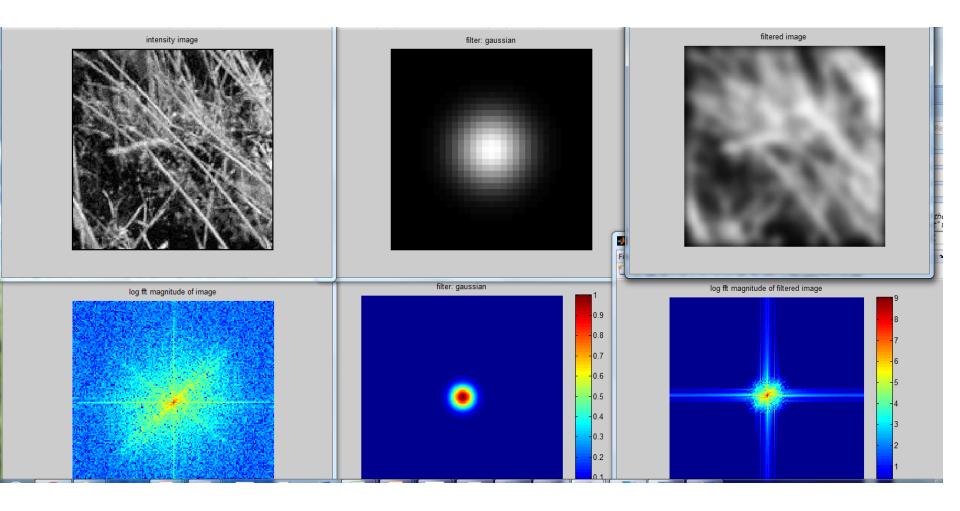
## Fourier Matlab demo

Filtering

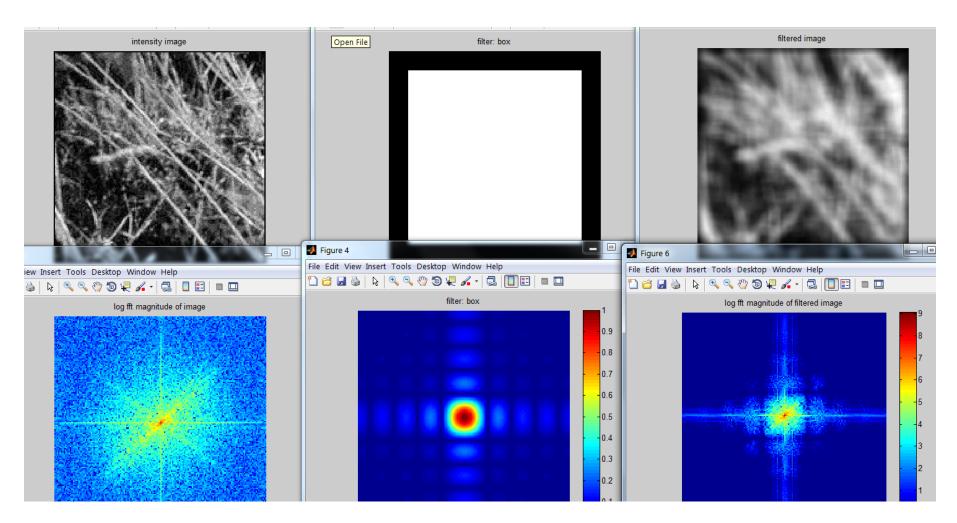
Why does the Gaussian give a nice smooth image, but the square filter give edgy artifacts?



#### Gaussian



#### **Box Filter**



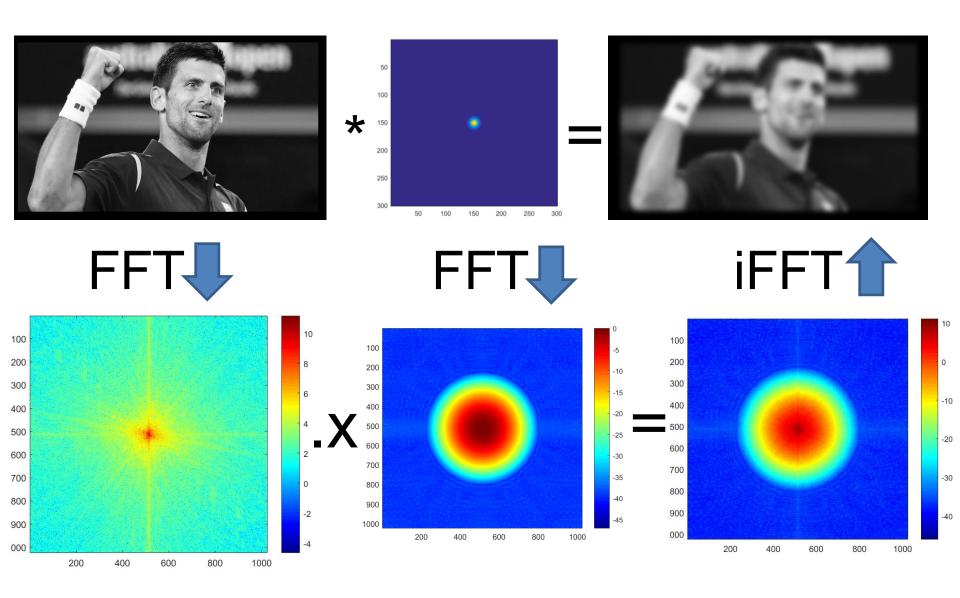
#### Is convolution invertible?

- If convolution is just multiplication in the Fourier domain, isn't deconvolution just division?
- Sometimes, it clearly is invertible (e.g. a convolution with an identity filter)
- In one case, it clearly isn't invertible (e.g. convolution with an all zero filter)
- What about for common filters like a Gaussian?

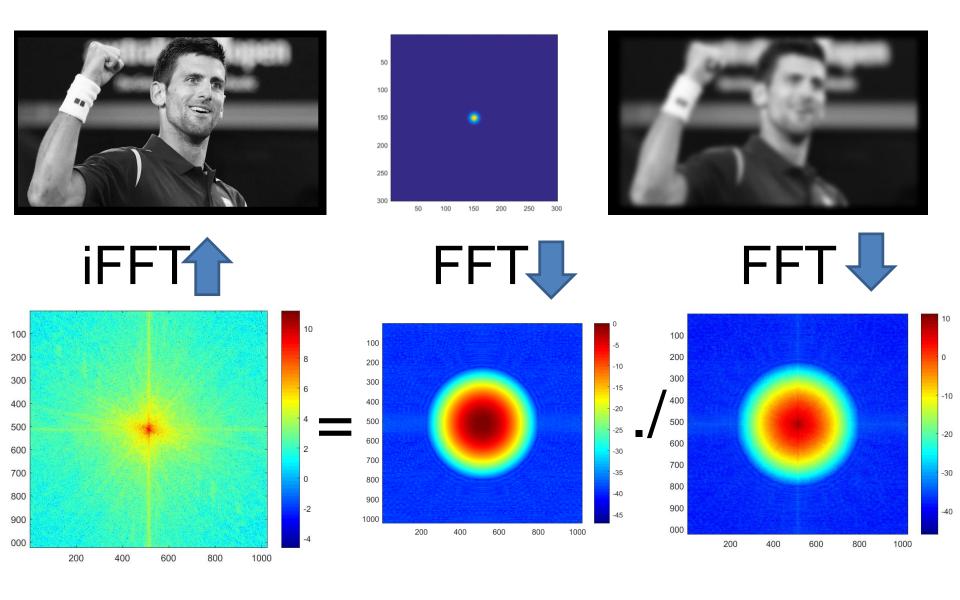
## Let's experiment on Novak



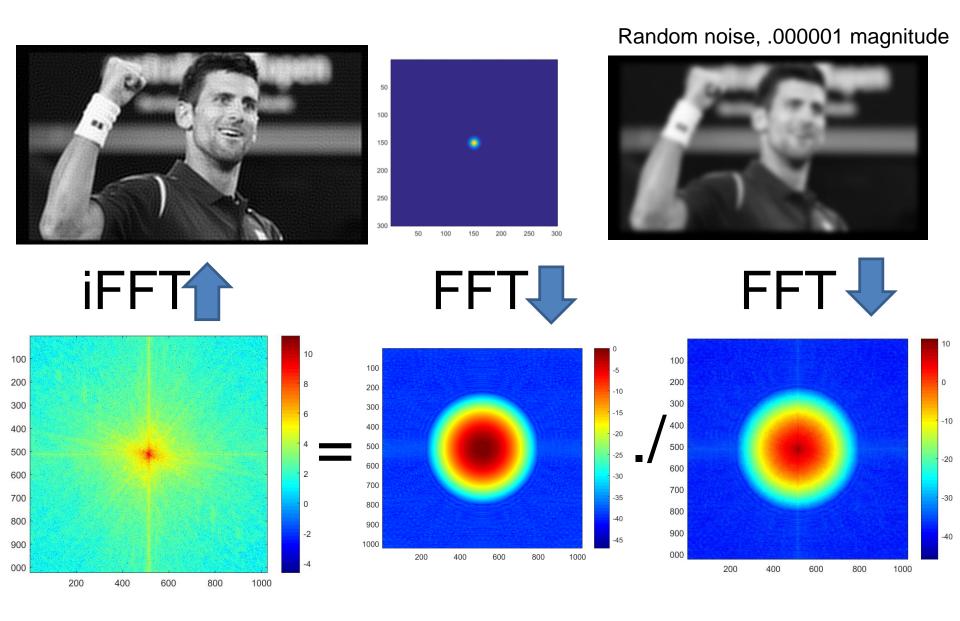
#### Convolution



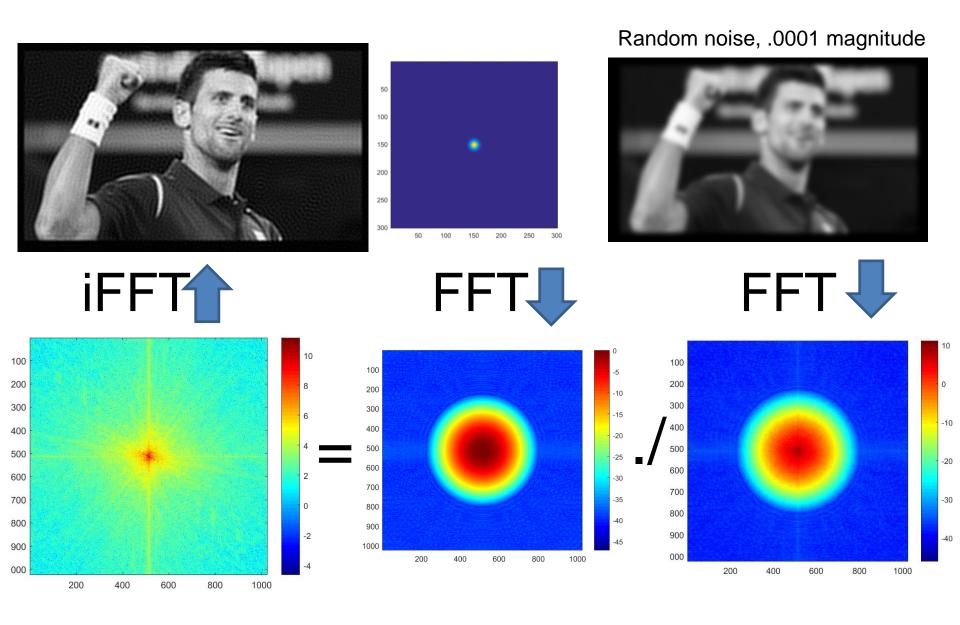
#### Deconvolution?



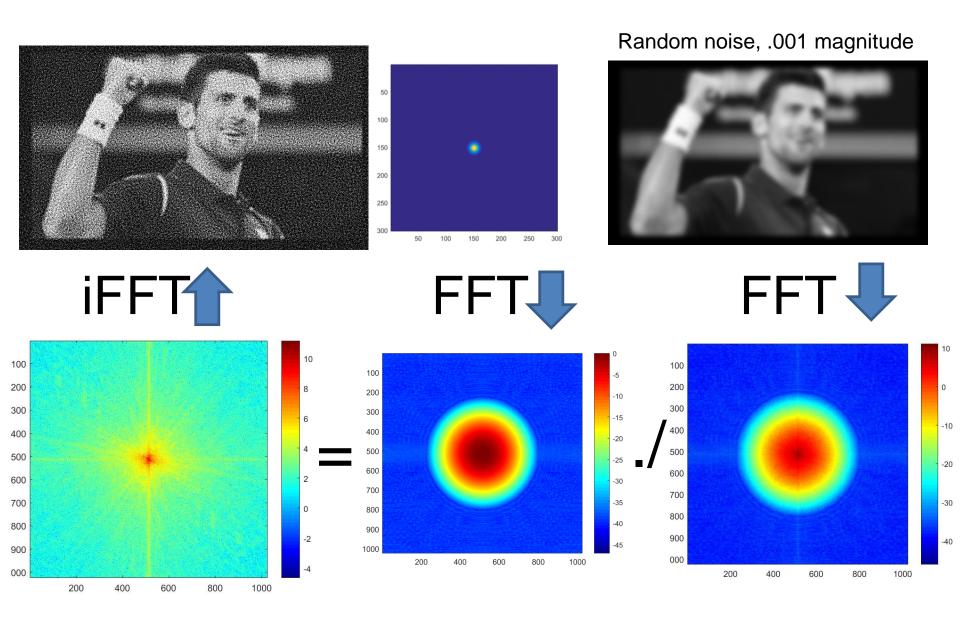
#### But under more realistic conditions



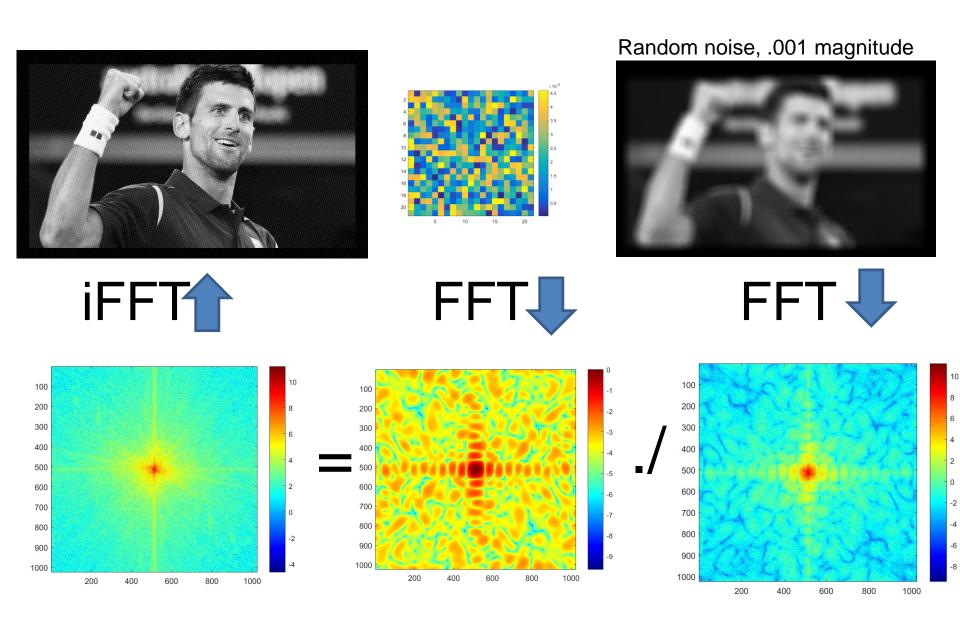
#### But under more realistic conditions



#### But under more realistic conditions



#### With a random filter...



#### Deconvolution is hard

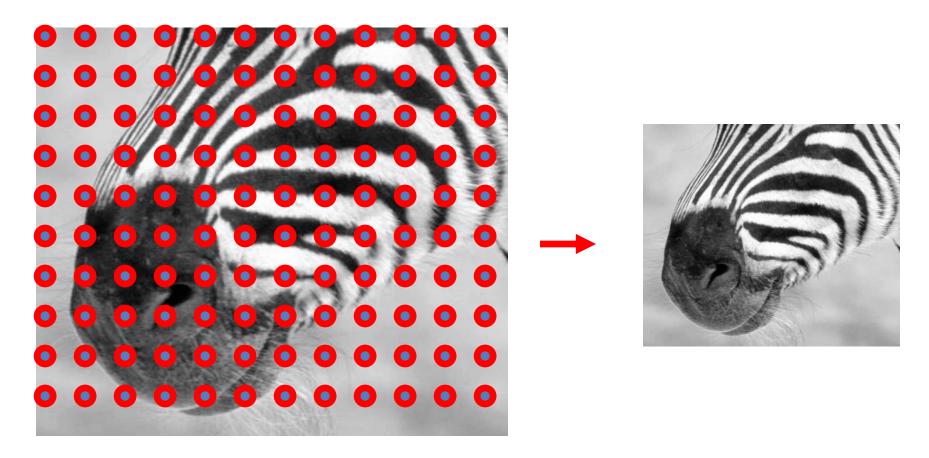
- Active research area.
- Even if you know the filter (non-blind deconvolution), it is still very hard and requires strong regularization.
- If you don't know the filter (blind deconvolution) it is harder still.

#### Sampling

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



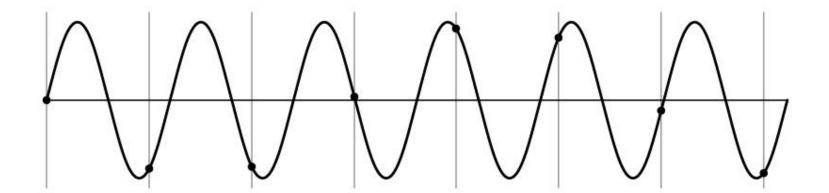
## Subsampling by a factor of 2



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

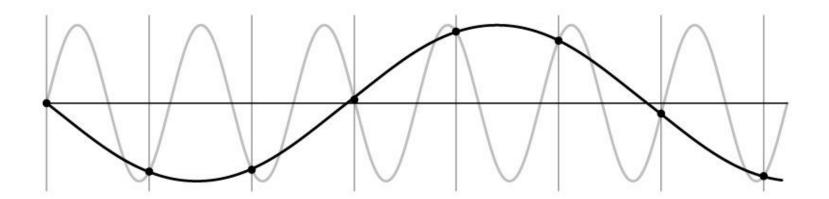
## Aliasing problem

1D example (sinewave):



## Aliasing problem

• 1D example (sinewave):



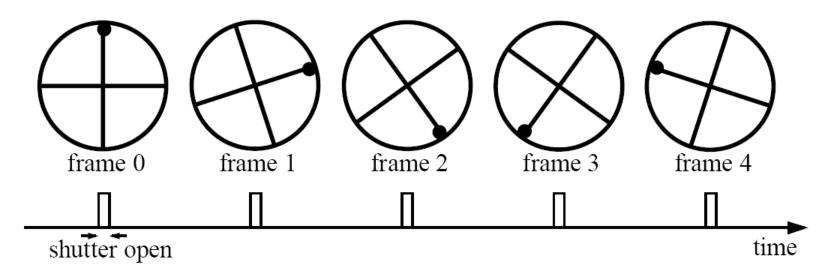
## Aliasing problem

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

## Aliasing in video

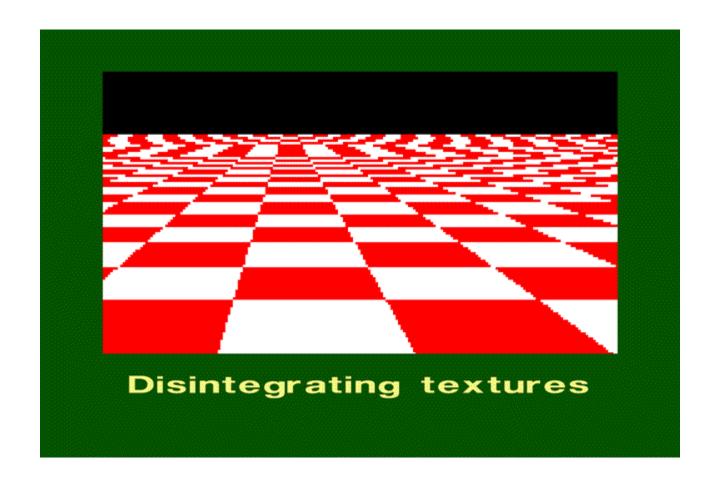
Imagine a spoked wheel moving to the right (rotating clockwise). Mark wheel with dot so we can see what's happening.

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):

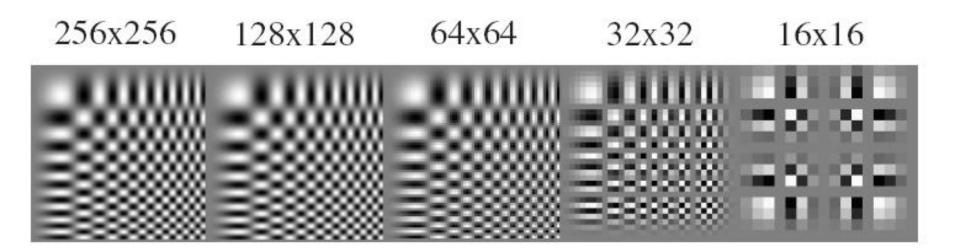


Without dot, wheel appears to be rotating slowly backwards! (counterclockwise)

## Aliasing in graphics

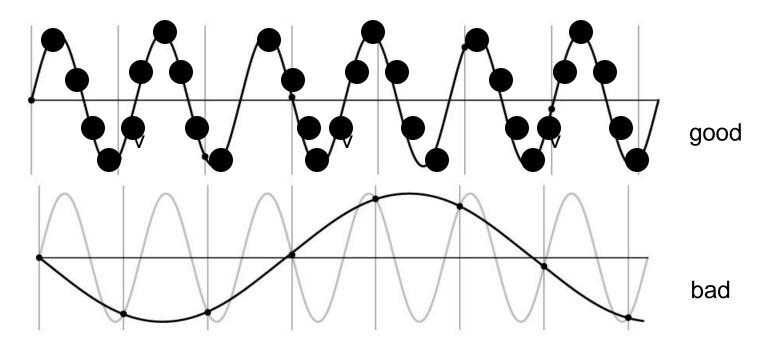


## Sampling and aliasing



## Nyquist-Shannon Sampling Theorem

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



## Anti-aliasing

#### Solutions:

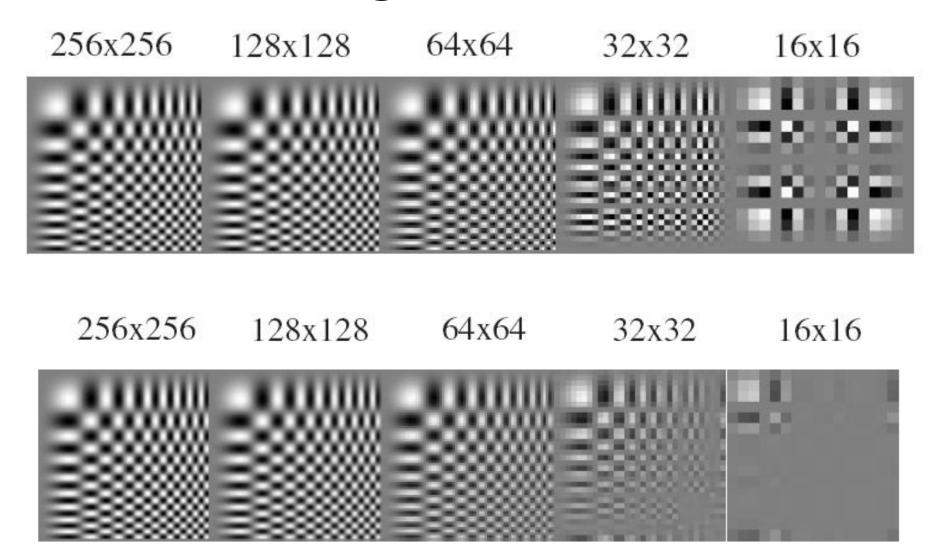
Sample more often

- 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

### 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



## Subsampling without pre-filtering



## Subsampling with Gaussian pre-filtering

