

---

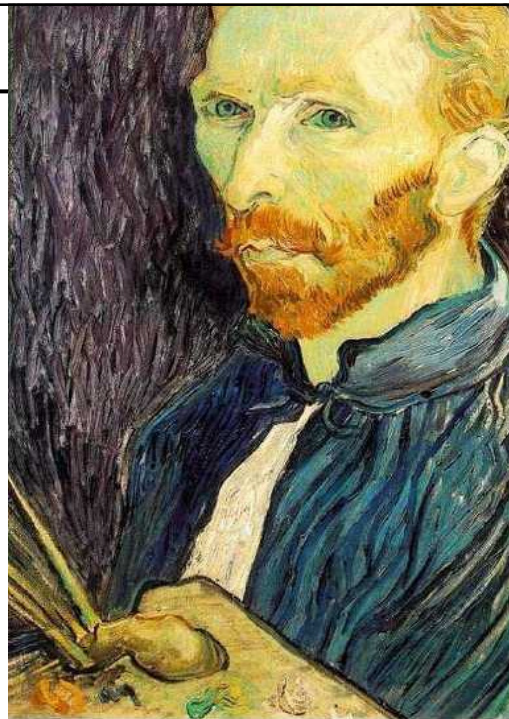
## Fourier Analysis

15-463: Rendering and Image  
Processing  
Alexei Efros

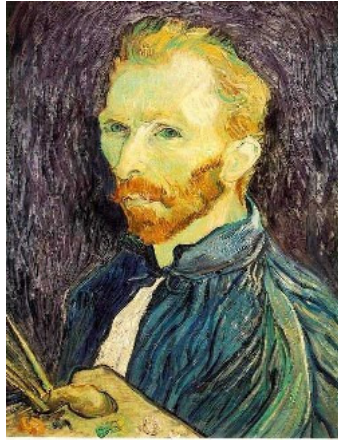
### Image Scaling

This image is too big to  
fit on the screen. How  
can we reduce it?

How to generate a half-  
sized version?



## Image sub-sampling



1/4

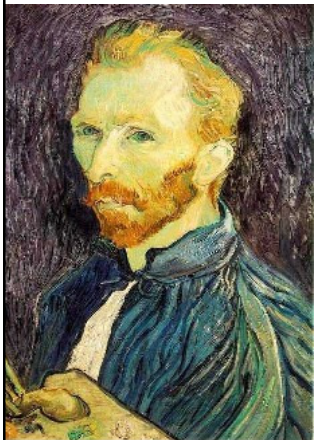


1/8

Throw away every other row and column to create a 1/2 size image  
- called *image sub-sampling*

Slide by Steve Seitz

## Image sub-sampling



1/2



1/4 (2x zoom)



1/8 (4x zoom)

Why does this look so cruffy?

Slide by Steve Seitz

## Even worse for synthetic images



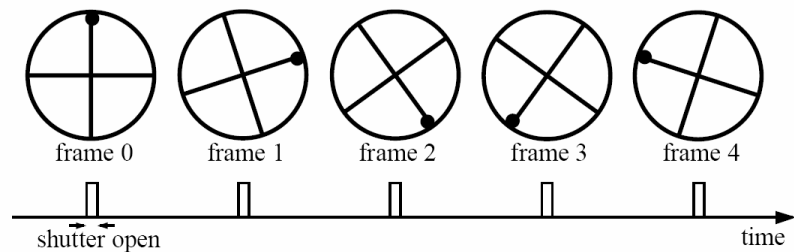
Slide by Steve Seitz

## Really bad in video

Imagine a spoked wheel moving to the right (rotating clockwise).

Mark wheel with dot so we can see what's happening.

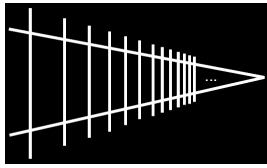
If camera shutter is only open for a fraction of a frame time (frame time =  $1/30$  sec. for video,  $1/24$  sec. for film):



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

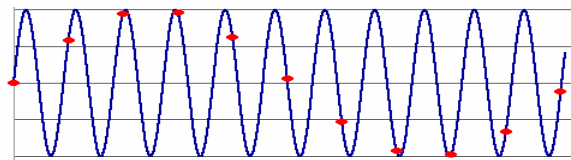
Slide by Paul Heckbert

## Alias: n., an assumed name



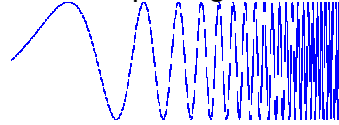
Picket fence receding  
Into the distance will  
produce aliasing...

WHY?

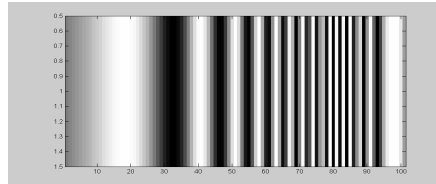


Not enough samples

Input signal:



Matlab output:



`x = 0:.05:5; imagesc(sin((2.^x).*x))`

Aj-aj-aj:  
Alias!

## Aliasing

- occurs when your sampling rate is not high enough to capture the amount of detail in your image
- Can give you the wrong signal/image—an *alias*

Where can it happen in graphics?

- During image synthesis:
  - **sampling** continuous signal into discrete signal
  - e.g. ray tracing, line drawing, function plotting, etc.
- During image processing:
  - **resampling** discrete signal at a different rate
  - e.g. Image warping, zooming in, zooming out, etc.

To do sampling right, need to understand the structure of  
your signal/image

Enter Monsieur Fourier...

## Jean Baptiste Fourier (1768-1830)

had crazy idea (1807):

**Any periodic 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 true!

- called Fourier Series



## A sum of sines

Our building block:

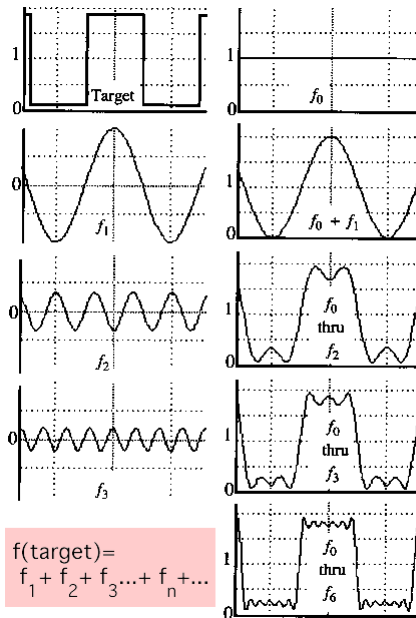
$$A \sin(\omega x + \phi)$$

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

How many degrees of freedom?

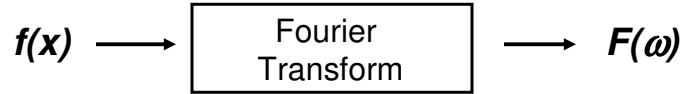
What does each control?

Which one encodes the coarse vs. fine structure of the signal?



## Fourier Transform

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



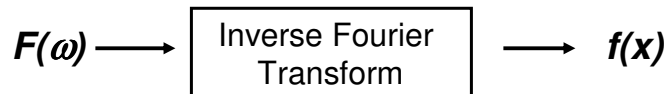
For every  $\omega$  from 0 to inf,  $F(\omega)$  holds the amplitude  $A$  and phase  $\phi$  of the corresponding sine  $A\sin(\omega x + \phi)$

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

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

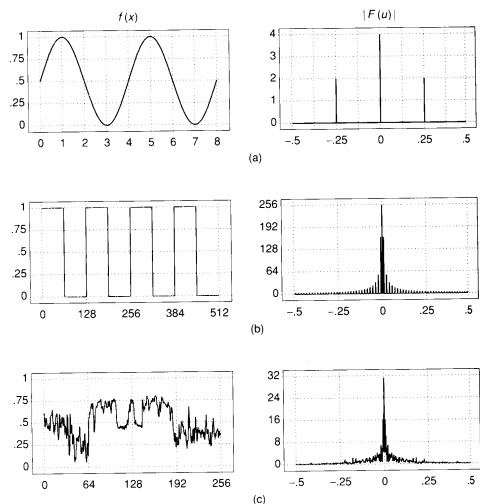
$$A = \pm\sqrt{R(\omega)^2 + I(\omega)^2} \qquad \phi = \tan^{-1} \frac{I(\omega)}{R(\omega)}$$

We can always go back:



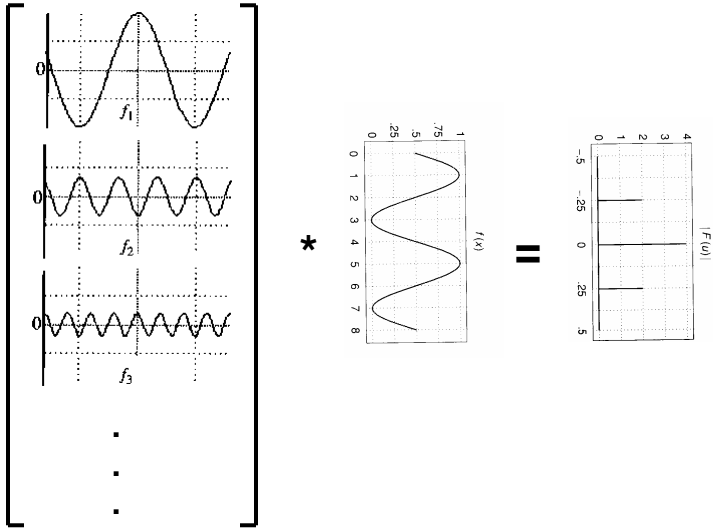
## Frequency Spectra

Usually, amplitude is more interesting than phase:



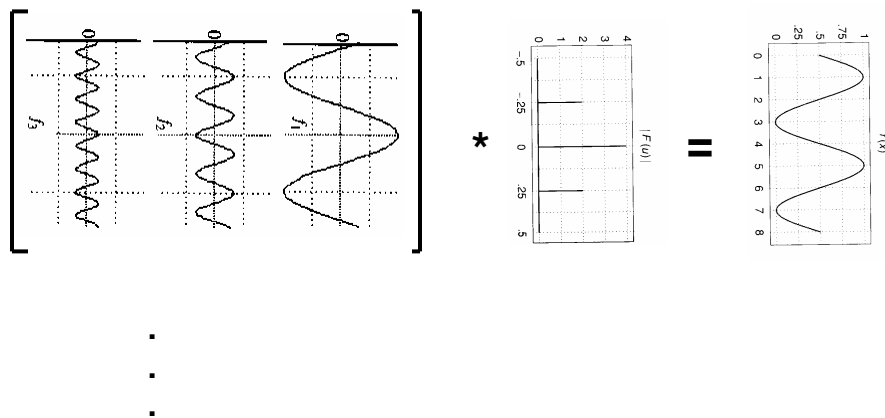
## FT: Just a change of basis

$$M * f(x) = F(\omega)$$



## IFT: Just a change of basis

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



## Finally: Scary Math

---

$$\text{Fourier Transform : } F(\omega) = \int_{-\infty}^{+\infty} f(x)e^{-i\omega x} dx$$

$$\text{Inverse Fourier Transform : } f(x) = \frac{1}{2\pi} \int_{-\infty}^{+\infty} F(\omega)e^{i\omega x} d\omega$$

## Finally: Scary Math

---

$$\text{Fourier Transform : } F(\omega) = \int_{-\infty}^{+\infty} f(x)e^{-i\omega x} dx$$

$$\text{Inverse Fourier Transform : } f(x) = \frac{1}{2\pi} \int_{-\infty}^{+\infty} F(\omega)e^{i\omega x} d\omega$$

...not really scary:  $e^{i\omega x} = \cos(\omega x) + i \sin(\omega x)$

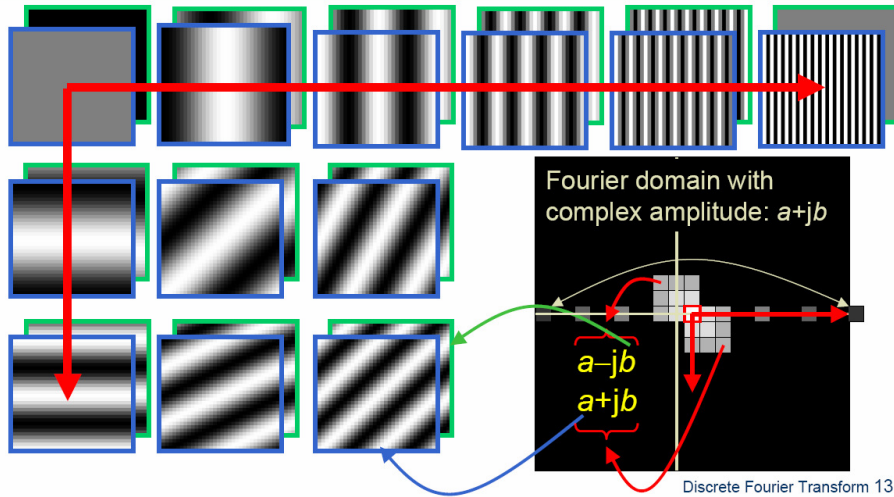
is hiding our old friend:  $A \sin(\omega x + \phi)$

$$\begin{array}{l} \text{phase can be encoded} \\ \text{by sin/cos pair} \end{array} \rightarrow \begin{array}{l} P \cos(x) + Q \sin(x) = A \sin(x + \phi) \\ A = \pm \sqrt{P^2 + Q^2} \quad \phi = \tan^{-1}\left(\frac{P}{Q}\right) \end{array}$$

So it's just our signal  $f(x)$  times sine at frequency  $\omega$

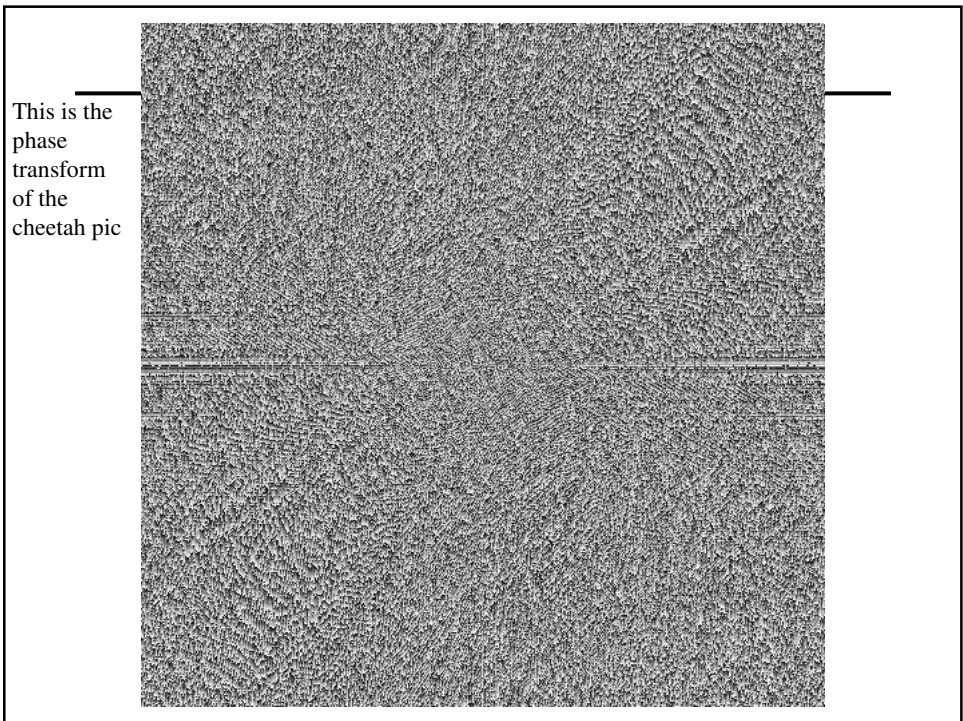
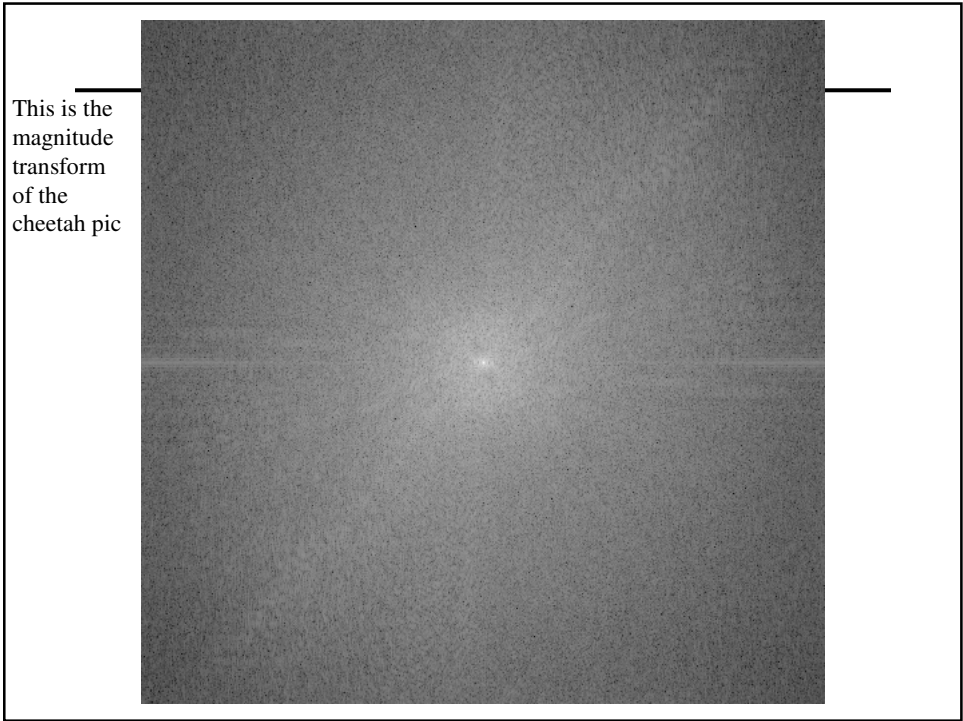


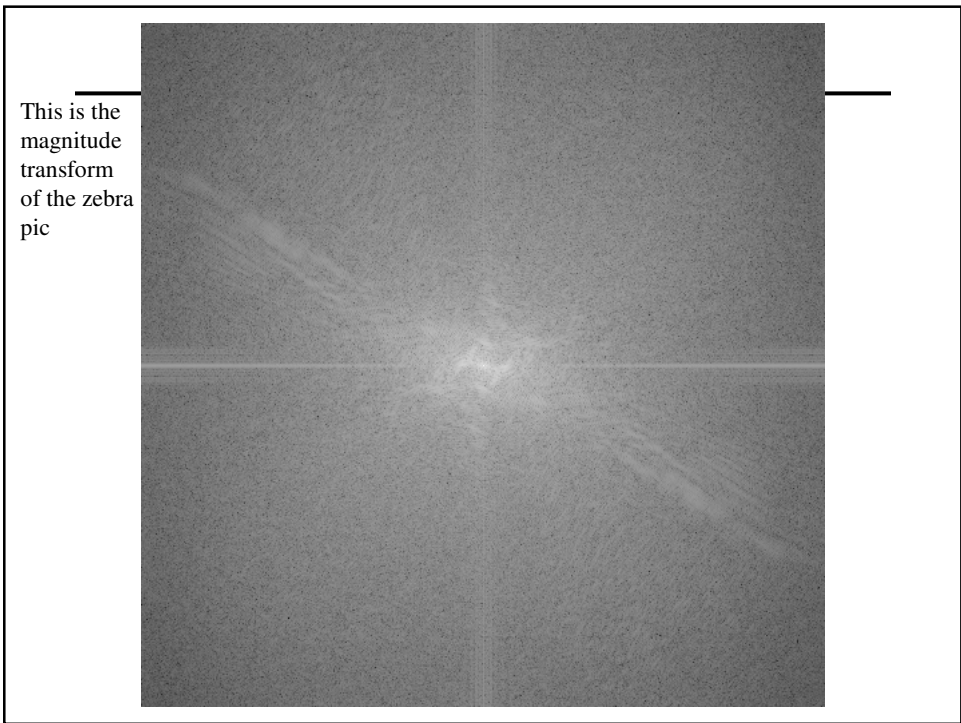
## Extension to 2D

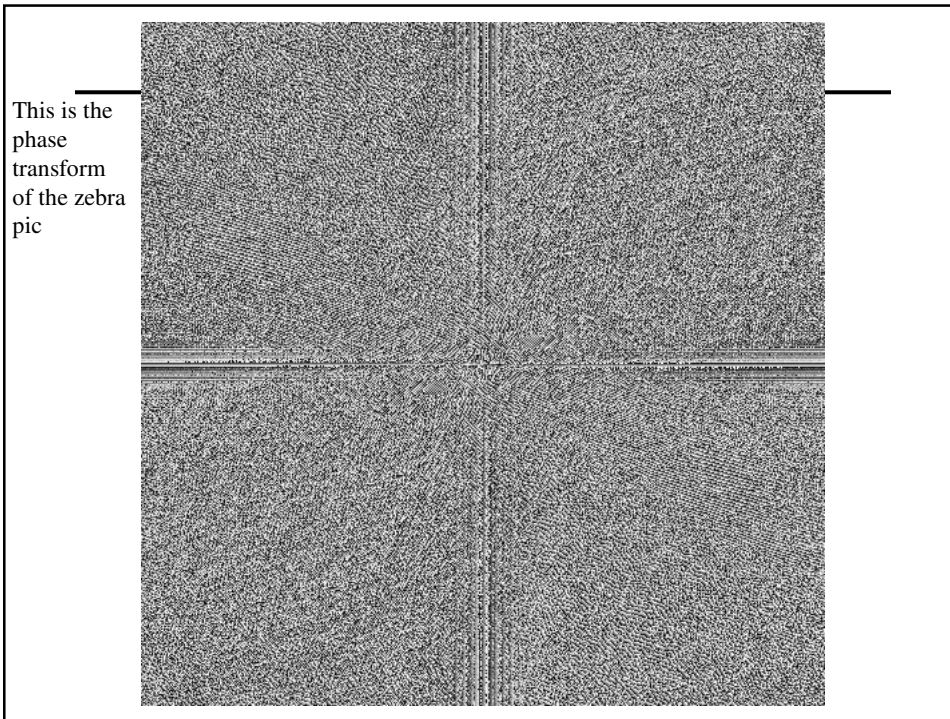


in Matlab, check out: `imagesc(log(abs(fftshift(fft2(im)))));`









## Curious things about FT on images

The magnitude spectra of all natural images quite similar

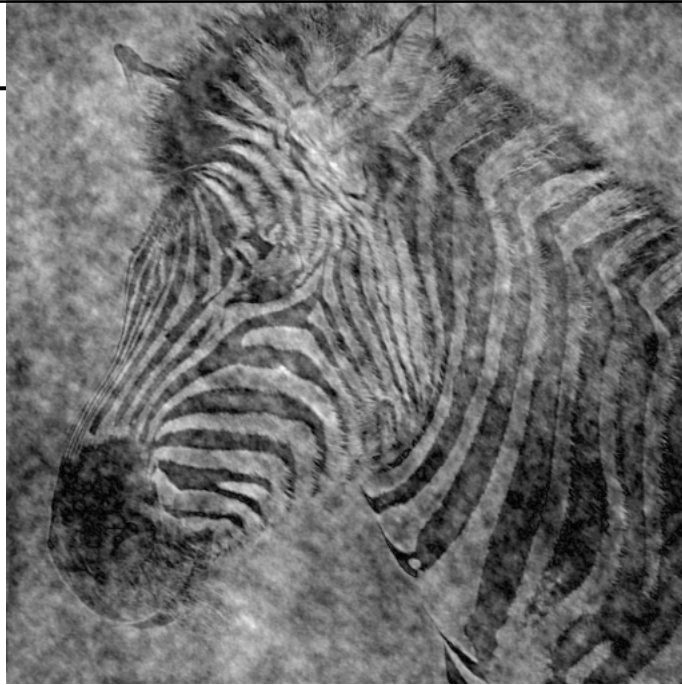
- Heavy on low-frequencies, falling off in high frequencies
- Will any image be like that, or is it a property of the world we live in?

Most information in the image is carried in the phase, not the amplitude

- Seems to be a fact of life
- Not quite clear why

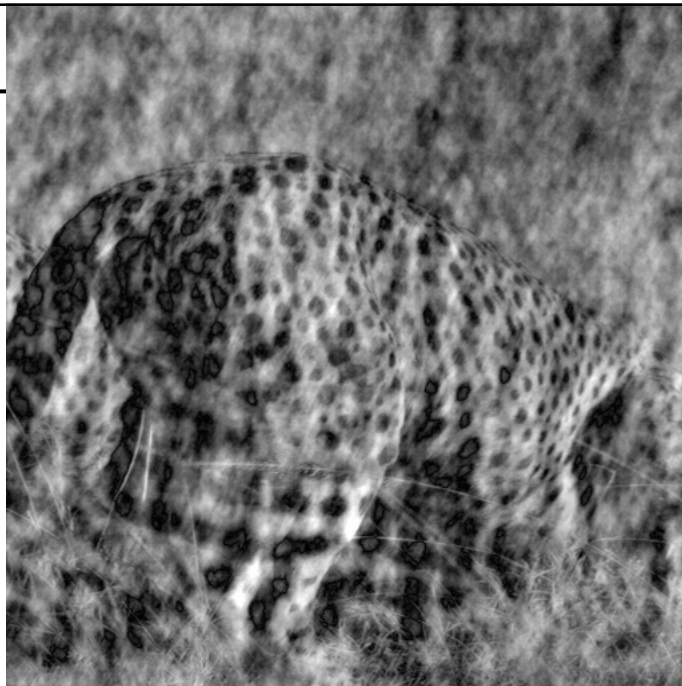
---

Reconstruction  
with zebra  
phase, cheetah  
magnitude



---

Reconstruction  
with cheetah  
phase, zebra  
magnitude



## Various Fourier Transform Pairs

### Important facts

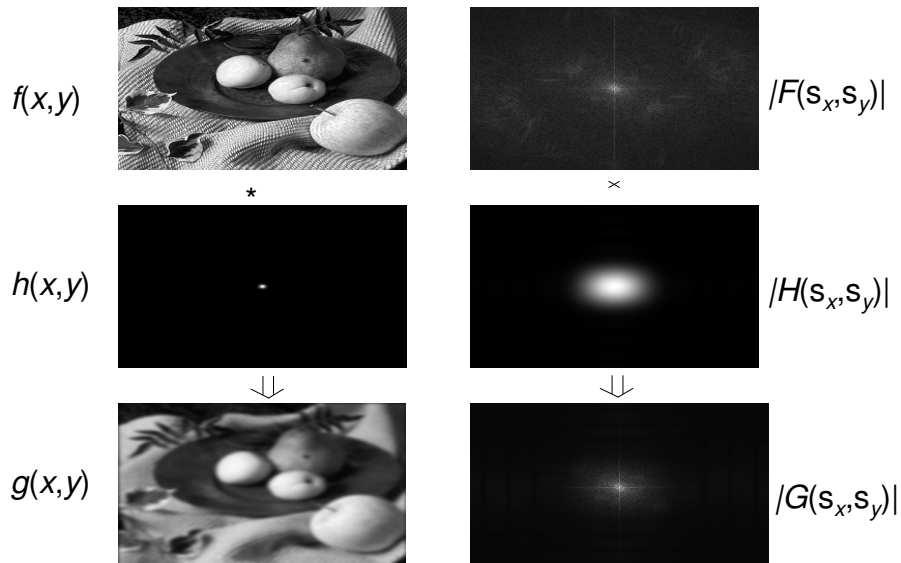
- The Fourier transform is linear
- There is an inverse FT
- if you scale the function's argument, then the transform's argument scales the other way. This makes sense --- if you multiply a function's argument by a number that is larger than one, you are stretching the function, so that high frequencies go to low frequencies
- The FT of a Gaussian is a Gaussian.

### The convolution theorem

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

Slide by David Forsyth

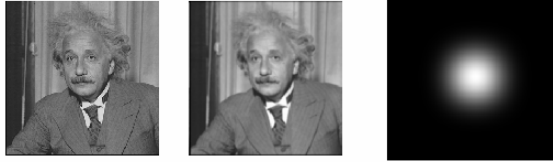
## 2D convolution theorem example



Slide by Steve Seitz

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

low-pass:



band-pass:



what's high-pass?