Fourier Transform and Frequency Domain

Hi, Dr. Elizabeth? Yeah, Uh... I accidentally took the Fourier transform of my cat... Meow

15-463, 15-663, 15-862 Computational Photography Fall 2017, Lecture 6

http://graphics.cs.cmu.edu/courses/15-463

Course announcements

- Last call for responses to Doodle about rescheduling the September 27th lecture!
 - Link available on Piazza.
 - Currently 17 responses. I'll pick a date on Tuesday evening.
- Homework 1 is being graded.
 - Grades with comments will be uploaded on Canvas hopefully by Wednesday.
 How was it?
- Homework 2 has been posted.
 - Much larger than homework 1.
 - Start early! Experiments take a long time to run.
 - How many have read/started/finished it?

Overview of today's lecture

- Some history.
- Fourier series.
- Frequency domain.
- Fourier transform.
- Frequency-domain filtering.
- Revisiting sampling.

Slide credits

Most of these slides were adapted from:

• Kris Kitani (15-463, Fall 2016).

Some slides were inspired or taken from:

- Fredo Durand (MIT).
- James Hays (Georgia Tech).

Some history

Who is this guy?



What is he famous for?



Jean Baptiste Joseph Fourier (1768-1830)

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Jean Baptiste Joseph Fourier (1768-1830) The Fourier series claim (1807): *Any univariate function can be rewritten as a*

weighted sum of sines and cosines of different frequencies.'

... and apparently also for the discovery of the greenhouse effect

Is this claim true?



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Is this claim true?



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Well, almost.

- The theorem requires additional conditions.
- Close enough to be named after him.
- Very surprising result at the time.

Is this claim true?



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The Fourier series claim (1807):

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The committee examining his paper had expressed skepticism, in part due to not so rigorous proofs

Malus

Lagrange

Legendre

Amusing aside



Only known portrait of Adrien-Marie Legendre

1820 watercolor <u>caricatures</u> of French mathematicians <u>Adrien-</u> <u>Marie Legendre</u> (left) and Joseph Fourier (right) by French artist <u>Julien-Leopold Boilly</u>

> For two hundred years, people were misidentifying this portrait as him



Louis Legendre (same last name, different person)

Fourier series

Basic building block

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

Fourier's claim: Add enough of these to get <u>any periodic</u> signal you want!

Basic building block



Fourier's claim: Add enough of these to get <u>any periodic</u> signal you want!



How would you generate this function?



 $\sin(2\pi x)$

















How would could you visualize this in the frequency domain?



Frequency domain



Recall the temporal domain visualization



Recall the temporal domain visualization



amplitude 1.00 0.66 0.33 0 k 2k 3k 4k frequency

Recall the temporal domain visualization










Spatial domain visualization



Frequency domain visualization

?



 k_x

Spatial domain visualization



Frequency domain visualization



 k_x

How would you generate this image with sine waves?



How would you generate this image with sine waves?



Has both an x and y components







Basic building block



Fourier's claim: Add enough of these to get <u>any periodic</u> signal you want!

Complex numbers have two parts:

rectangular coordinates
$$R+jI_{
m what's\,this?}$$
 what's this?



330°

Complex numbers have two parts:





330°

Complex numbers have two parts:



Alternative reparameterization:

polar coordinates

$$r(\cos heta+j\sin heta)$$
 how do we compute these?



polar transform

Complex numbers have two parts:

rectangular coordinates R+jI real imaginary

Alternative reparameterization:

polar coordinates

$$r(\cos heta + j \sin heta)$$

polar transform
 $heta = an^{-1}(rac{I}{R})$ $r = \sqrt{R^2 + I^2}$



polar transform

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polar transform
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polar transform

How do you write these in exponential form?





This will help us understand the Fourier transform equations



Where is the connection to the 'summation of sine waves' idea?



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Where is the connection to the 'summation of sine waves' idea?

$$f(x) = \sum_{k=0}^{N-1} F(k) e^{j2\pi kx/N}$$
Euler's formula
$$e^{j\theta} = \cos \theta + j \sin \theta$$
sum over frequencies
$$f(x) = \sum_{k=0}^{N-1} F(k) \left\{ \cos(2\pi kx) + j \sin(2\pi kx) \right\}$$
scaling parameter
wave components

Fourier transform pairs



Computing the discrete Fourier transform (DFT)

Computing the discrete Fourier transform (DFT)

 $F(k) = \frac{1}{N} \sum_{x=0}^{N-1} f(x) e^{-j2\pi kx/N}$ is just a matrix multiplication:

F = Wf

$$\begin{bmatrix} F(0) \\ F(1) \\ F(2) \\ F(3) \\ \vdots \\ F(N-1) \end{bmatrix} = \begin{bmatrix} W^0 & W^0 & W^0 & W^0 & \cdots & W^0 \\ W^0 & W^1 & W^2 & W^3 & \cdots & W^{N-1} \\ W^0 & W^2 & W^4 & W^6 & \cdots & W^{N-2} \\ W^0 & W^3 & W^6 & W^9 & \cdots & W^{N-3} \\ \vdots & & & \ddots & \vdots \\ W^0 & W^{N-1} & W^{N-2} & W^{N-3} & \cdots & W^1 \end{bmatrix} \begin{bmatrix} f(0) \\ f(1) \\ f(2) \\ f(3) \\ \vdots \\ f(N-1) \end{bmatrix} \qquad W = e^{-j2\pi/N}$$

In practice this is implemented using the *fast Fourier transform* (FFT) algorithm.

Another way to compute the Fourier transform

Use a lens!



An ideal thin lens is an optical Fourier transform engine.



Fourier transforms of natural images



original



amplitude





phase

Fourier transforms of natural images

Image phase matters!



cheetah phase with zebra amplitude



zebra phase with cheetah amplitude

Frequency-domain filtering

Why do we care about all this?

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!

What do we use convolution for?

Convolution for 1D continuous signals

Definition of linear shift-invariant filtering as convolution:

$$(f*g)(x) = \int_{-\infty}^{\infty} f(y)g(x-y)dy$$
filter signal

Using the convolution theorem, we can interpret and implement all types of linear shift-invariant filtering as multiplication in frequency domain.

Why implement convolution in frequency domain?

Frequency-domain filtering in Matlab

Filtering with fft:

```
im = double(imread('...'))/255;
im = rgb2gray(im); % "im" should be a gray-scale floating point image
[imh, imw] = size(im);
hs = 50; % filter half-size
fil = fspecial('gaussian', hs*2+1, 10);
fftsize = 1024; % should be order of 2 (for speed) and include padding
im fft = fft2(im, fftsize, fftsize);
                                                          % 1) fft im with
padding
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

Spatial domain filtering



Frequency domain filtering

Revisiting blurring

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



Gaussian blur













Box blur










A lens' kernel is its aperture

This is (one of the reasons) why we try to make lens apertures as circular as possible.



An ideal thin lens is an optical Fourier transform engine.



circular aperture (Airy disk)



rectangular aperture







filters shown in frequencydomain



filters shown in frequencydomain

?



high-pass



high-pass



original image



low-pass filter





original image



low-pass filter







original image



high-pass filter





original image



high-pass filter







original image



band-pass filter







original image



band-pass filter







original image



band-pass filter







original image



band-pass filter







Revisiting sampling

The Nyquist-Shannon sampling theorem

A continuous signal can be perfectly reconstructed from its discrete version if sampling occurred with frequency:

$$f_s \geq 2 f_{\max} \quad \longleftarrow \quad {}^{ ext{This is called the}}_{ ext{Nyquist frequency}}$$

Equivalent reformulation: When downsampling, aliasing does not occur if samples are taken at the Nyquist frequency or higher.

The Nyquist-Shannon sampling theorem

A continuous signal can be perfectly reconstructed from its discrete version if sampling occurred with frequency:

 $f_{\rm s} \ge 2 f_{\rm max} \quad \longleftarrow \quad {\rm This \ is \ called \ the} {\rm Nyquist \ frequency}$

This is called the

Gaussian pyramid



How does the Nyquist-Shannon theorem relate to the Gaussian pyramid?

Gaussian pyramid



How does the Nyquist-Shannon theorem relate to the Gaussian pyramid?

- Gaussian blurring is low-pass filtering.
- By blurring we try to sufficiently decrease the Nyquist frequency to avoid aliasing.

How large should the Gauss blur we use be?

Frequency-domain filtering in human vision



Gala Contemplating the Mediterranean Sea Which at Twenty Meters Becomes the Portrait of Abraham Lincoln (Homage to Rothko)

Salvador Dali, 1976

Frequency-domain filtering in human vision



Low-pass filtered version

Frequency-domain filtering in human vision



High-pass filtered version

Variable frequency sensitivity

Experiment: Where do you see the stripes?



contrast

Variable frequency sensitivity

Campbell-Robson contrast sensitivity curve



contrast

- Early processing in humans filters for various orientations and scales of frequency
- Perceptual cues in the mid frequencies dominate perception

frequency

References

Basic reading:

• Szeliski textbook, Sections 3.4.

Additional reading:

- Goodman, "Introduction to Fourier Optics," W.H.Freeman Publishing 2004. the standard reference on Fourier optics
- Hubel and Wiesel, "Receptive fields, binocular interaction and functional architecture in the cat's visual cortex," The Journal of Physiology 1962

a foundational paper describing information processing in the visual system, including the different types of filtering it performs; Hubel and Wiesel won the Nobel Prize in Medicine in 1981 for the discoveries described in this paper