Frequency

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- Fourier transform
- Sampling and Aliasing
- 3 Hybrid Images

Input for slides includes content by Steve Seitz, Trevor Darrell, James Hays, Kristen Grauman, Antonio Torralba, Li Fei Fei. David Jacobs, Derek Hoiem

Jean Baptiste Joseph Fourier (1768-1830)

had crazy idea (1807):

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

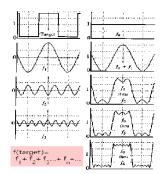
- · Don't believe it?
 - Neither did Lagrange, Laplace, Poisson and other big wigs
 - Not translated into English until 1878!
- But it's (mostly) true!
 - called Fourier Series
 - there are some subtle restrictions

...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.



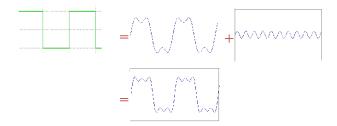
Our building block: $A\sin(\omega x + \phi)$

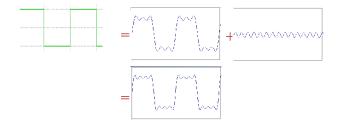
Add enough of them to get any signal g(x) you want!

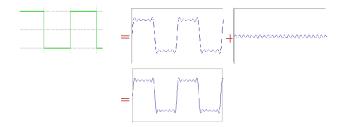


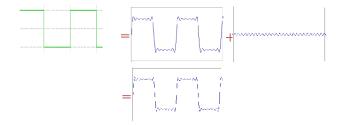
• example : $g(t) = \sin(2\pi f t) + (1/3)\sin(2\pi(3f) t)$ t) 0.3 0.

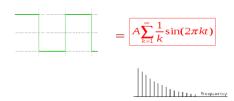












Example in image



Fourier I mage

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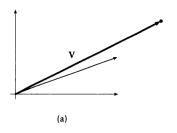
http://sharp.bu.edu/~slehar/fourier/fourier.html#filtering

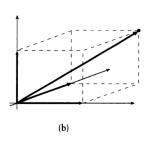
Example in image



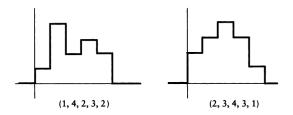
http://sharp.bu.edu/~slehar/fourier/fourier.html#filtering More: http://www.cs.unm.edu/~brayer/vision/fourier.html

Projection of points in Space





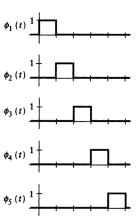
Projection of Functions



How to form the function in terms of basis functions?

$$f(t) = c_1 \phi_1(t) + c_2 \phi_2(t) + \dots + c_n \phi_n(t)$$
 (1)

Projection of Functions



Fourier transform

In the same way we obtain Fourier transform as projection of a signal h(x) on to a sinusoidal basis function In the continuous domain it is given by

$$H(\omega) = \int_{-\infty}^{\infty} h(x)e^{(-j\omega x)}dx \tag{2}$$

Illustration of Fourier transform

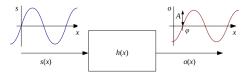


Figure 3.24 The Fourier Transform as the response of a filter h(x) to an input sinusoid $s(x) = e^{j\omega x}$ yielding an output sinusoid $o(x) = h(x) * s(x) = Ae^{j\omega x + \phi}$.

The Fourier transform is a tabulation of the magnitude and phase response at each frequency

$$H(\omega) = F\{h(x)\} = Ae^{j\phi} \tag{3}$$

where A is the magnitude and ϕ is the phase response The magnitude encodes how much signal there is at a particular frequency. The phase encodes spatial information indirectly.

Discrete Fourier transform

Now, in the discrete domain we have samples only at discrete intervals, i.e. h(x) = h[1], h[2], ..., h[n]

Therefore, for discrete signals, we have

$$H\omega = \int_0^{(N-1)T} h(x)e^{-j\omega x} dx \tag{4}$$

$$H\omega = h[0]e^{-j0} + h[1]e^{-j\omega T} + \dots + f[k]e^{-j\omega kT}$$
 (5)

And therefore the Fourier transform in discrete domain is given by

$$H(\omega) = \sum_{0}^{N-1} h(x)e^{-j\omega kT}$$
 (6)

Convolution property of Fourier Transform

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

$$F[g*h] = F[g]F[h] \tag{7}$$

 Convolution in spatial domain is equivalent to multiplication in frequency domain

$$g * h = F^{-1}[F[g]F[h]]$$
 (8)

Filtering in spatial domain

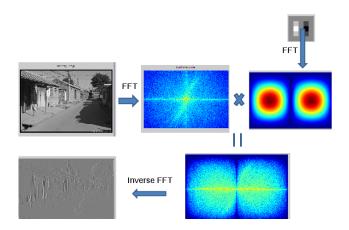








Filtering in frequency domain



FFT 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
```

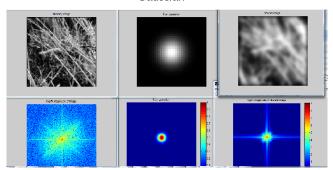
Filtering

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



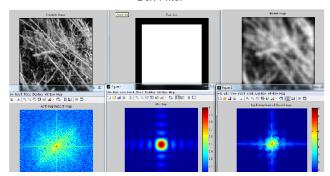
Gaussian Filter

Gaussian



Box Filter

Box Filter

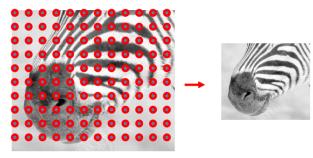


Sampling

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



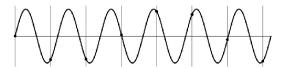
Subsampling by a factor of 2



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

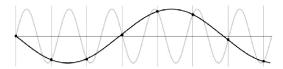
Aliasing Problem

• 1D example (sinewave):



Aliasing Problem

• 1D example (sinewave):



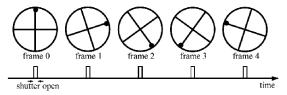
Aliasing Problem

- 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.

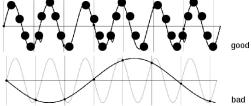
If camera shutter is only open for a fraction of a frame time (frame time = 1/30 sec. for video, 1/24 sec. for film):



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

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 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

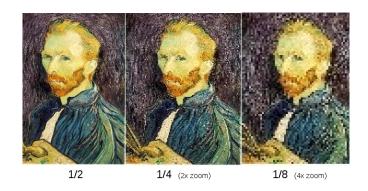
- 1. Start with image(h, w)
- 2. Apply low-pass filter

```
im_blur = imfilter(image, fspecial('gaussian', 7, \overline{1}))
```

3. Sample every other pixel

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

Sub-sampling without pre-filtering



Sub-sampling with pre-filtering



Salvador Dali invented Hybrid Images?



Salvador Dali



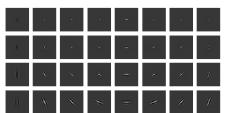


 A. Oliva, A. Torralba, P.G. Schyns, "Hybrid Images," SIGGRAPH 2006



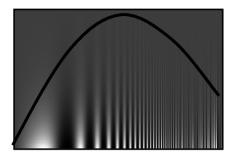
Clues from Human perception

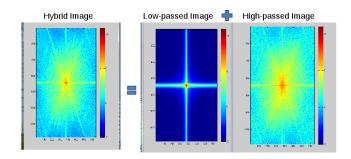
- Early processing in humans filters for various orientations and scales of frequency
- Perceptual cues in the mid-high frequencies dominate perception
- When we see an image from far away, we are effectively subsampling it



Early Visual Processing: Multi-scale edge and blob filters

Campbell-Robson contrast sensitivity curve





Example of Hybrid Images

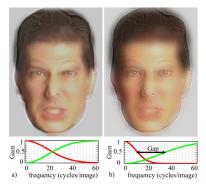


Figure 5: An angry man or a thoughtful woman? Both hybrid images are produced by combining the faces of an angry man (low spatial frequencies) and a stern woman (high spatial frequencies). You can switch the percept by watching the picture from a few meters. a) Bad hybrid image. The image looks ambiguous from up close due to the filter overlap. b) Good Hybrid image.

Example of Hybrid Images

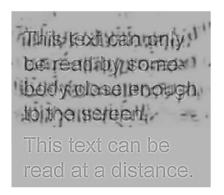


Figure 8: The hybrid font becomes invisible at few meters. The bottom text remains easy to read at relatively long distances.

Recap

- Fourier transform obtained by projection of signal onto sinusoidal signals
- Sampling of images may result in aliasing and anti-aliasing by following Nyquist theorem for sampling frequency
- Application of frequency domain transform for creating Hybrid images