

Signal Processing

COS 323

Digital “Signals”

- 1D: functions of space or time (e.g., sound)
- 2D: often functions of 2 spatial dimensions (e.g. images)
- 3D: functions of 3 spatial dimensions (CAT, MRI scans) or 2 space, 1 time (video)

Digital Signal Processing

1. Understand analogues of *filters*
2. Understand nature of *sampling*

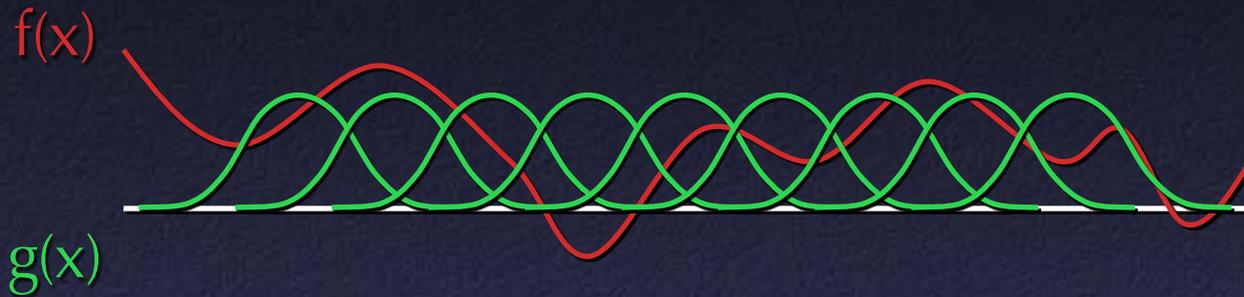
Filtering

- Consider a noisy 1D signal $f(x)$
- Basic operation: smooth the signal
 - Output = new function $h(x)$
 - Want properties: linearity, shift invariance
- Linear Shift-Invariant Filters
 - If you double input, double output
 - If you shift input, shift output

Convolution

- Output signal at each point = weighted average of local region of input signal
 - Depends on input signal, pattern of weights
 - “Filter” $g(x)$ = function of weights for linear combination
 - Basic operation = move filter to some position x , add up f times g

Convolution



$$f(x) * g(x) = \int_{-\infty}^{\infty} f(t) g(x-t) dt$$

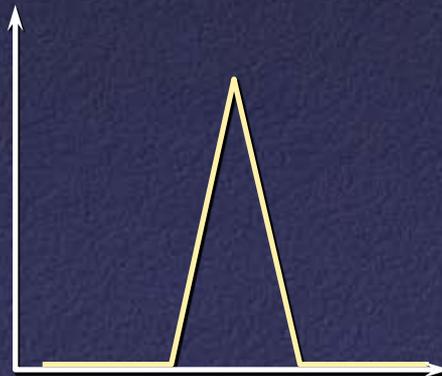
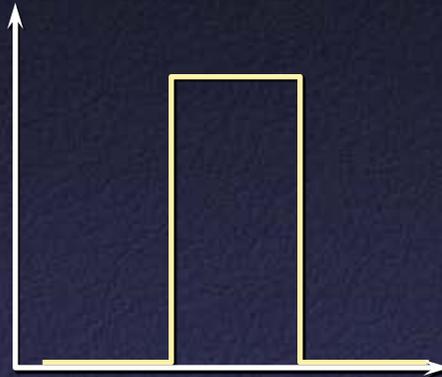
Convolution

- f is called “signal” and g is “filter” or “kernel”, but the operation is symmetric
- Usually desirable to leave a constant signal unchanged: choose g such that

$$\int_{-\infty}^{\infty} g(t) dt = 1$$

Filter Choices

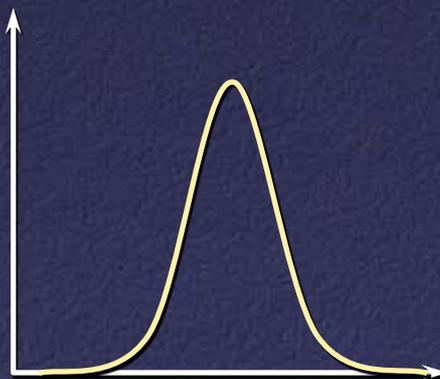
- Simple filters: box, triangle



Gaussian Filter

- Very commonly used filter

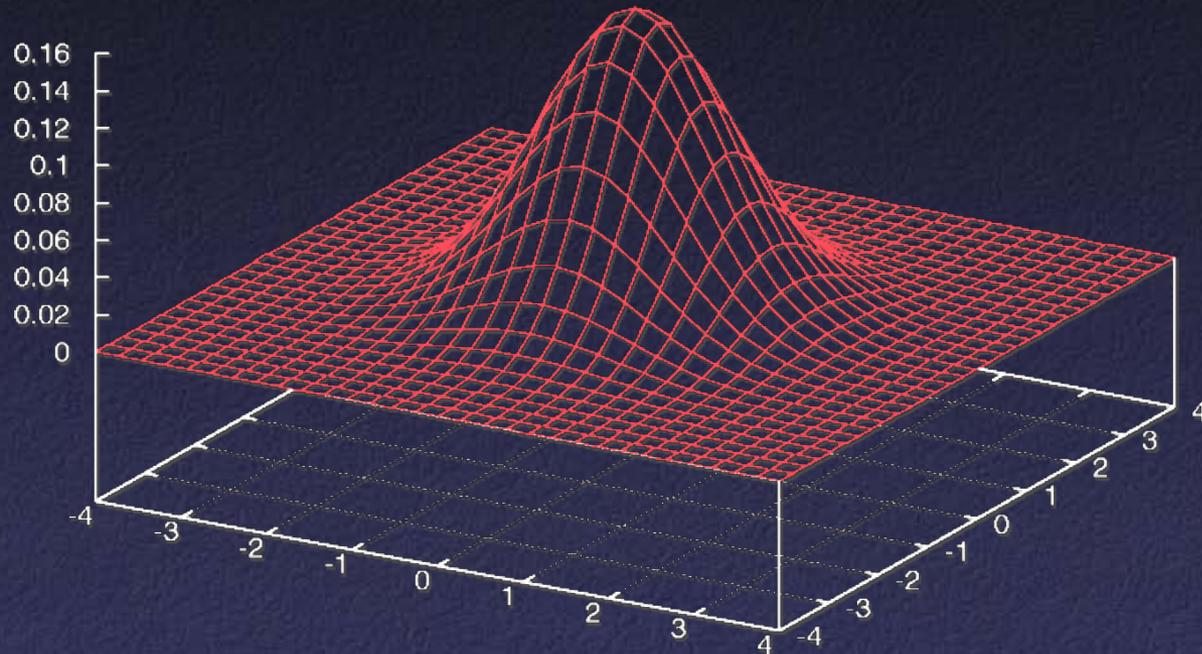
$$G(x) = \frac{1}{\sigma\sqrt{2\pi}} e^{-\frac{x^2}{2\sigma^2}}$$



Gaussian Filters

- Gaussians are used because:
 - Smooth (infinitely differentiable)
 - Decay to zero rapidly
 - Simple analytic formula
 - Separable: multidimensional Gaussian = product of Gaussians in each dimension
 - Convolution of 2 Gaussians = Gaussian
 - Limit of applying multiple filters (*) is Gaussian (Central limit theorem)

2D Gaussian Filter

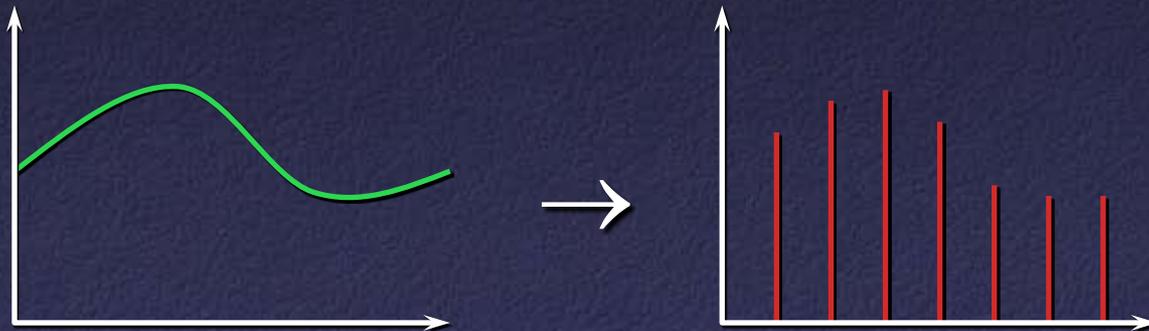


Sampled Signals

- Can't store continuous signal: instead store "samples"

- Usually evenly sampled:

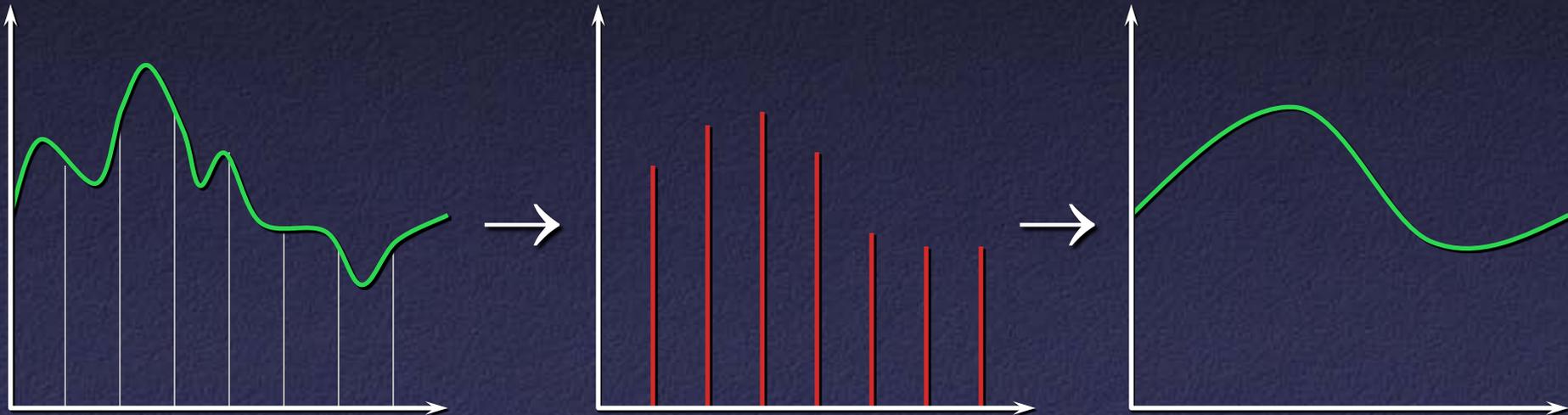
$$f_0 = f(x_0), f_1 = f(x_0 + \Delta x), f_2 = f(x_0 + 2\Delta x), f_3 = f(x_0 + 3\Delta x), \dots$$



- Instantaneous measurements of continuous signal
 - This can lead to problems

Aliasing

- Reconstructed signal might be very different from original: “aliasing”



- Solution: smooth the signal *before* sampling

Discrete Convolution

- Integral becomes sum over samples

$$f * g = \sum_i f_i g_{x-i}$$

- Normalization condition is

$$\sum_i g_i = 1$$

Computing Discrete Convolutions

$$f * g = \sum_i f_i g_{x-i}$$

- What happens near edges of signal?
 - Ignore (Output is smaller than input)
 - Pad with zeros (edges get dark)
 - Replicate edge samples
 - Wrap around
 - Reflect
 - Change filter

Computing Discrete Convolutions

$$f * g = \sum_i f_i g_{x-i}$$

- If f has n samples and g has m nonzero samples, straightforward computation takes time $O(nm)$
- OK for small filter kernels, bad for large ones

Example: Smoothing



Original image



Smoothed with
2D Gaussian kernel

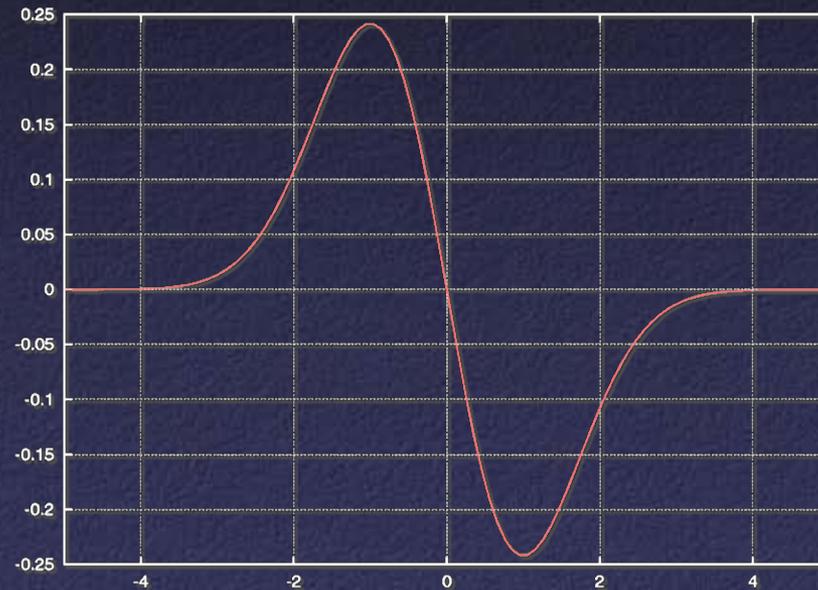
Example: Smoothed Derivative

- Derivative of noisy signal = more noisy
- Solution: smooth with a Gaussian *before* taking derivative
- Differentiation and convolution both linear operators: they “commute”

$$\frac{d}{dx}(f * g) = \frac{df}{dx} * g = f * \frac{dg}{dx}$$

Example: Smoothed Derivative

- Result: good way of finding derivative = convolution with derivative of Gaussian



Smoothed Derivative in 2D

- What is “derivative” in 2D? Gradient:

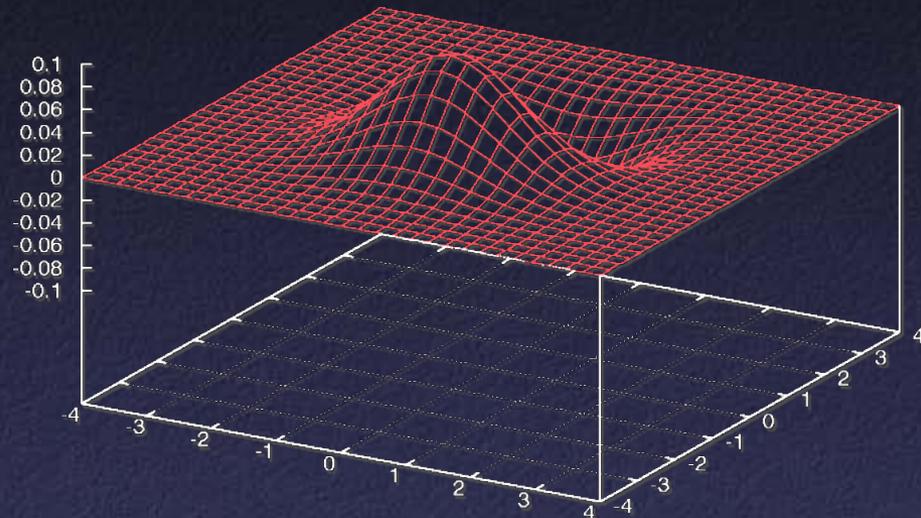
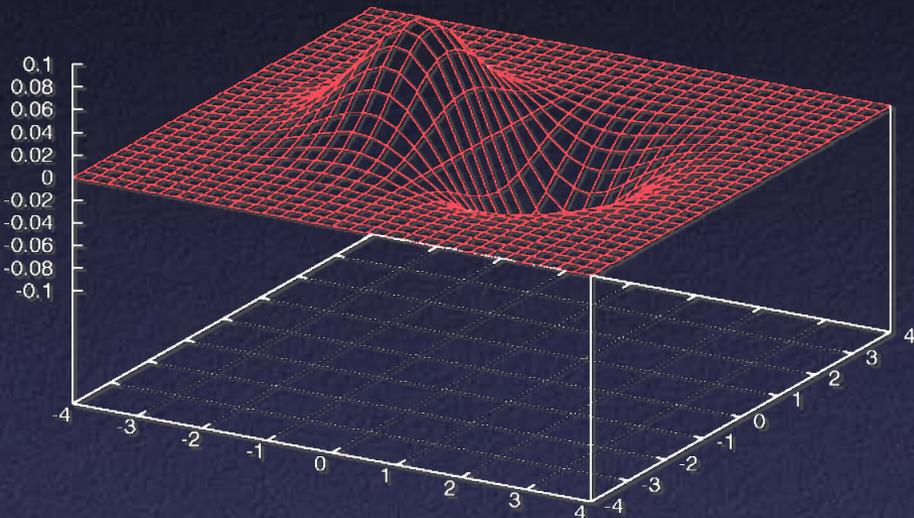
$$\nabla f(x, y) = \left(\frac{\partial f}{\partial x}, \frac{\partial f}{\partial y} \right)$$

- Gaussian is separable! $G_2(x, y) = G_1(x)G_1(y)$

- Combine smoothing, differentiation:

$$\nabla(f(x, y) * G_2(x, y)) = \begin{bmatrix} f(x, y) * (G_1'(x)G_1(y)) \\ f(x, y) * (G_1(x)G_1'(y)) \end{bmatrix} = \begin{bmatrix} f(x, y) * G_1'(x) * G_1(y) \\ f(x, y) * G_1(x) * G_1'(y) \end{bmatrix}$$

Smoothed Derivative in 2D



$$\nabla(f(x, y) * G_2(x, y)) = \begin{bmatrix} f(x, y) * (G_1'(x)G_1(y)) \\ f(x, y) * (G_1(x)G_1'(y)) \end{bmatrix} = \begin{bmatrix} f(x, y) * G_1'(x) * G_1(y) \\ f(x, y) * G_1(x) * G_1'(y) \end{bmatrix}$$

Smoothed Derivative in 2D



Original Image



Smoothed Gradient Magnitude

Canny Edge Detector

- Smooth
- Find derivative
- Find maxima
- Threshold

Canny Edge Detector



Original Image



Edges

Fourier Transform

- Transform applied to function to analyze its “frequency” content
- Several versions
 - Fourier series:
 - input = continuous, bounded; output = discrete, unbounded
 - Fourier transform:
 - input = continuous, unbounded; output = continuous, unbounded
 - Discrete Fourier transform (DFT):
 - input = discrete, bounded; output = discrete, bounded

Fourier Series

- Periodic function $f(x)$ defined over $[-\pi .. \pi]$

$$f(x) = \frac{1}{2}a_0 + \sum_{n=1}^{\infty} a_n \cos(nx) + b_n \sin(nx)$$

where

$$a_n = \frac{1}{\pi} \int_{-\pi}^{\pi} f(x) \cos(nx) dx$$

$$b_n = \frac{1}{\pi} \int_{-\pi}^{\pi} f(x) \sin(nx) dx$$

Fourier Series

- This works because sines, cosines are orthonormal over $[-\pi .. \pi]$:

$$\frac{1}{\pi} \int_{-\pi}^{\pi} \cos(mx) \cos(nx) dx = \delta_{mn}$$

$$\frac{1}{\pi} \int_{-\pi}^{\pi} \sin(mx) \sin(nx) dx = \delta_{mn}$$

$$\frac{1}{\pi} \int_{-\pi}^{\pi} \sin(mx) \cos(nx) dx = 0$$

- Kronecker delta:

$$\delta_{mn} = \begin{cases} 1 & \text{if } m = n \\ 0 & \text{otherwise} \end{cases}$$

Fourier Transform

- Continuous Fourier transform:

$$F(k) = \mathcal{F}(f(x)) = \int_{-\infty}^{\infty} f(x) e^{2\pi i k x} dx$$

- Discrete Fourier transform:

$$F_k = \sum_{x=0}^{n-1} f_x e^{2\pi i \frac{k}{n} x}$$

- F is a function of frequency – describes how much of each frequency f contains
- Fourier transform is invertible

Fourier Transform and Convolution

- Fourier transform turns convolution into multiplication:

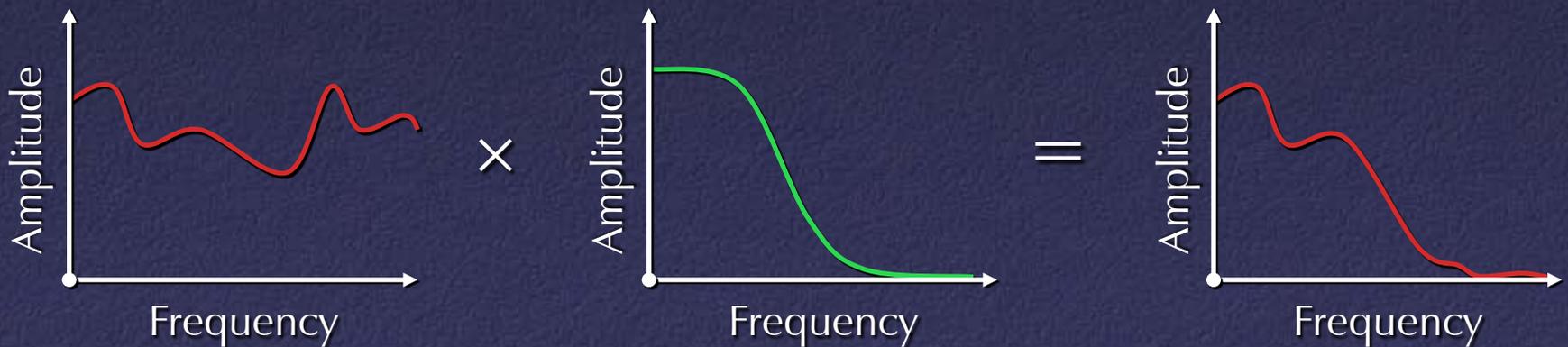
$$\mathcal{F}(f(x) * g(x)) = \mathcal{F}(f(x)) \mathcal{F}(g(x))$$

(and vice versa):

$$\mathcal{F}(f(x) g(x)) = \mathcal{F}(f(x)) * \mathcal{F}(g(x))$$

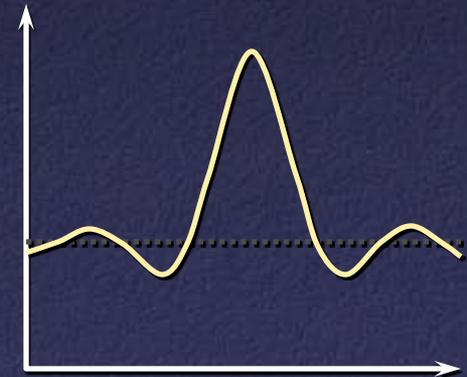
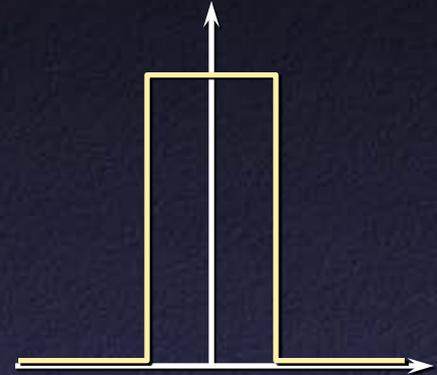
Fourier Transform and Convolution

- Useful application #1: Use frequency space to understand effects of filters
 - Example: Fourier transform of a Gaussian is a Gaussian
 - Thus: attenuates high frequencies



Fourier Transform and Convolution

- Box function?
- In frequency space:
sinc function
 - $\text{sinc}(x) = \sin(x) / x$
 - Not as good at attenuating high frequencies



Fourier Transform and Convolution

- Fourier transform of derivative:

$$\mathcal{F}\left(\frac{d}{dx} f(x)\right) = 2\pi i k \mathcal{F}(f(x))$$

- Blows up for high frequencies!
 - After Gaussian smoothing, doesn't blow up

Fourier Transform and Convolution

- Useful application #2: Efficient computation
 - Fast Fourier Transform (FFT) takes time
 $O(n \log n)$
 - Thus, convolution can be performed in time
 $O(n \log n + m \log m)$
 - Greatest efficiency gains for large filters