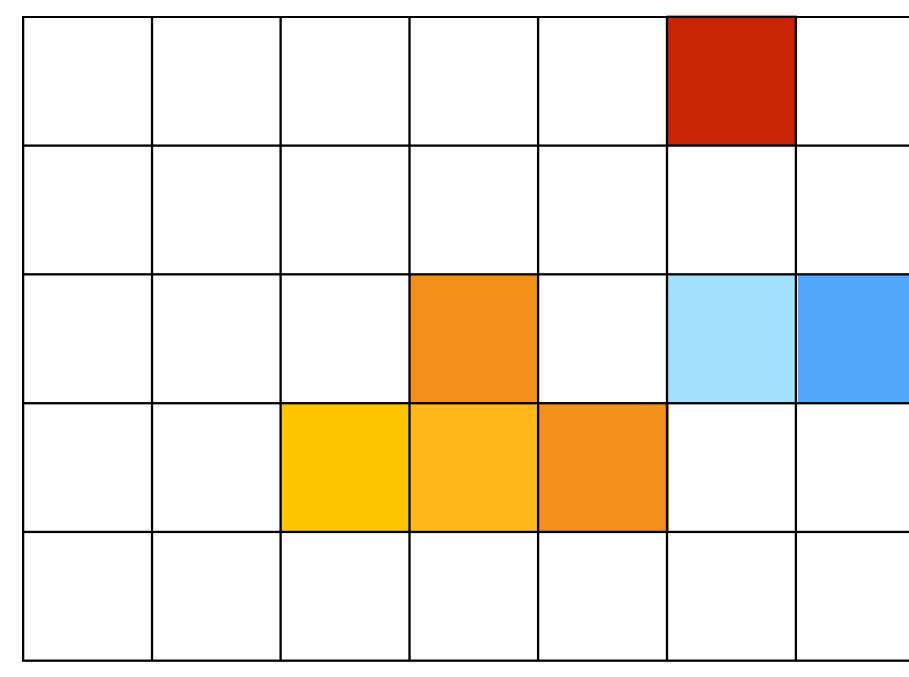
Lecture 2:

Drawing a Triangle (+ the basics of sampling and anti-aliasing)

Interactive Computer Graphics Stanford CS248, Winter 2021

Last time

- A very simple notion of digital image representation
 - that we are about to challenge!
- An image: a 2D array of color values



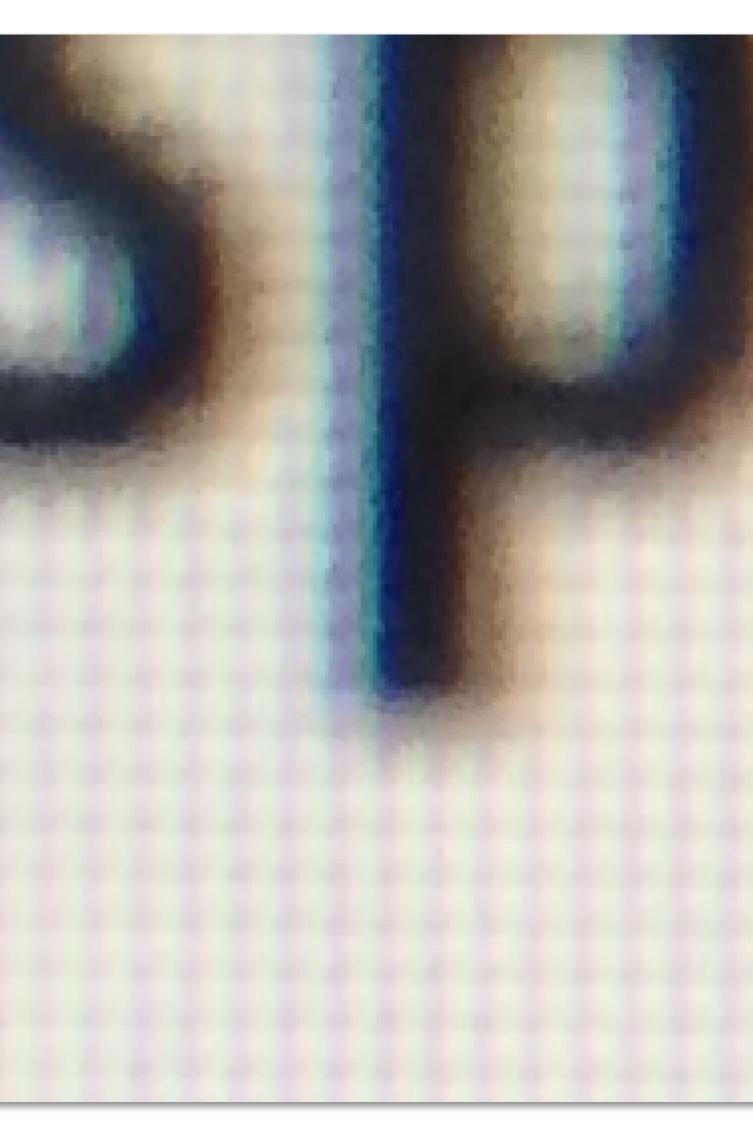




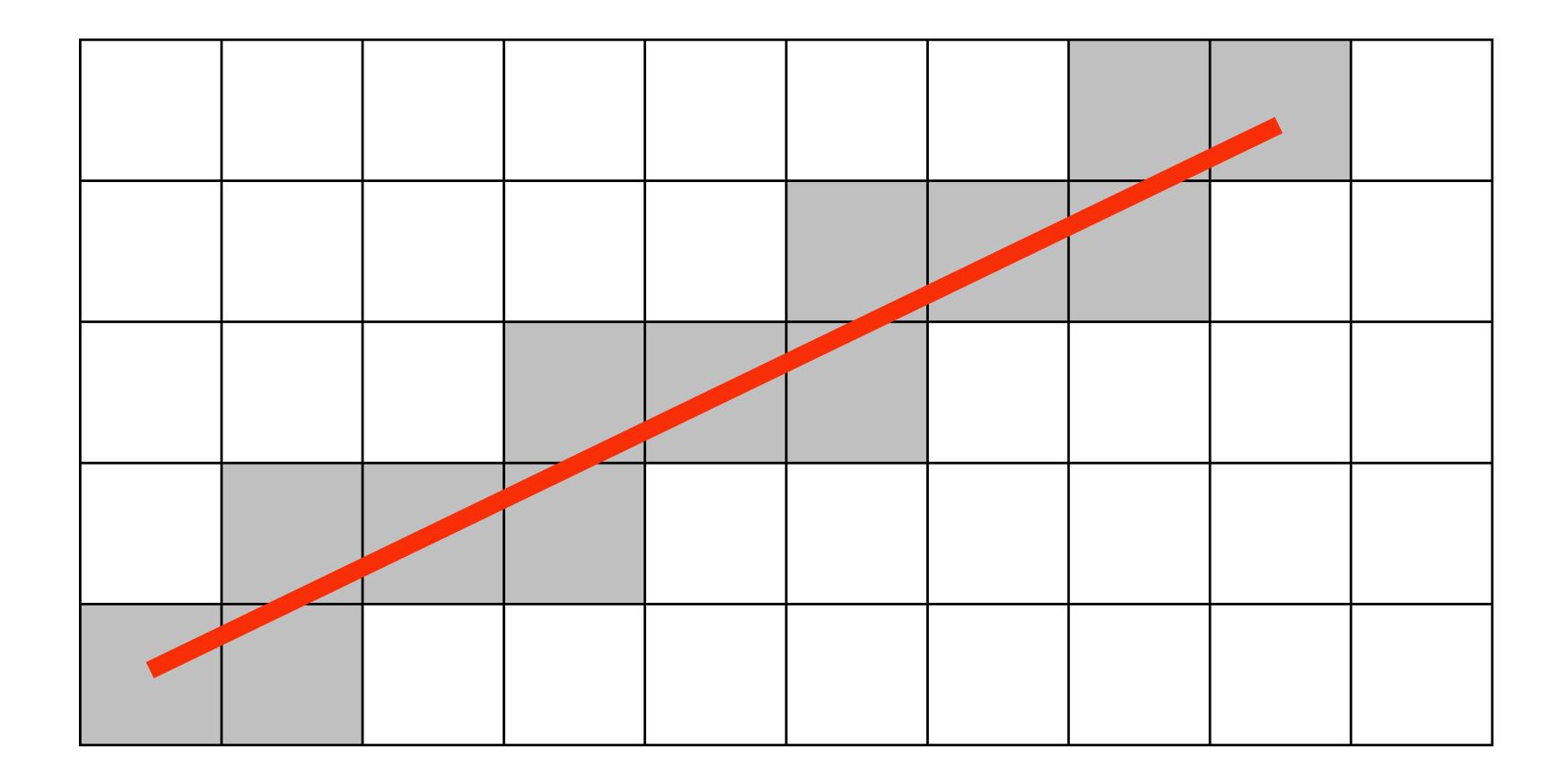
Last time

A display converts a color value at each pixel in an image to emitted light

Display pixel on my laptop

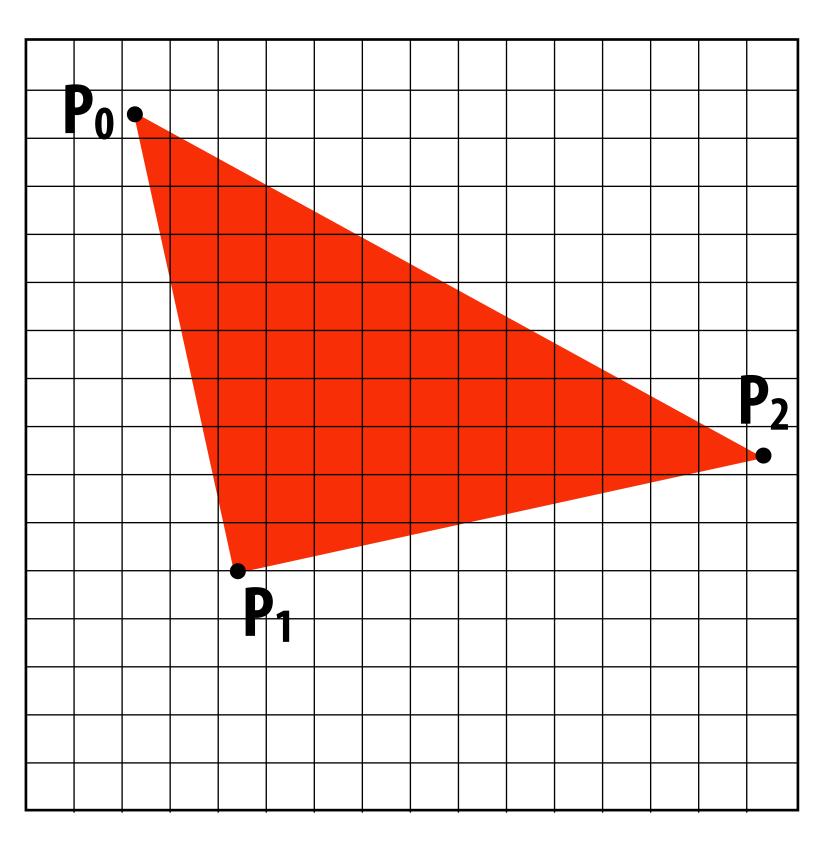


Last time: What pixels should we color in to draw a line?

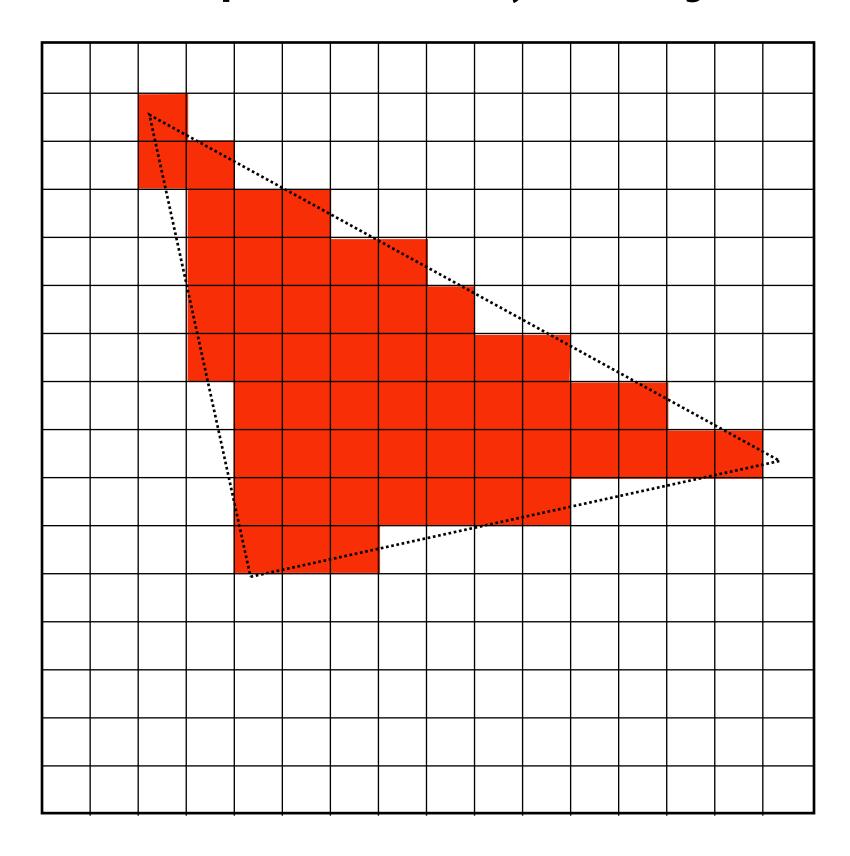


Today: drawing a triangle (Converting a representation of a triangle into an image) "Triangle rasterization"

Input: 2D position of triangle vertices: P₀, P₁, P₂



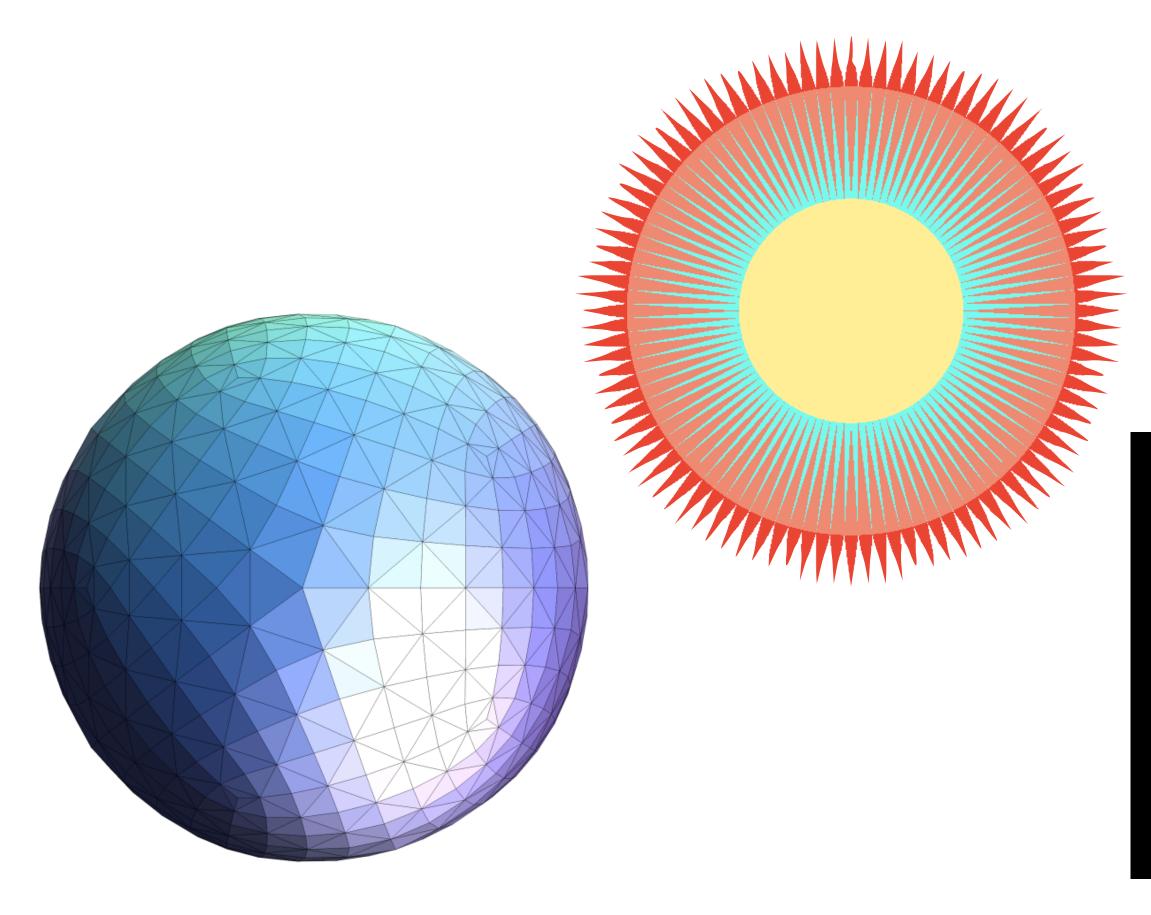


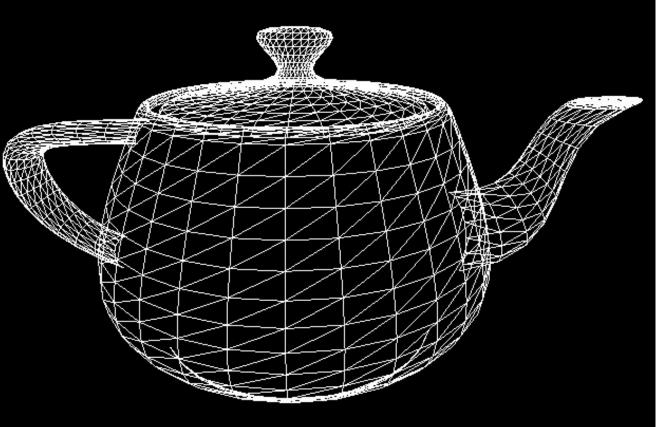


Output: set of pixels "covered" by the triangle

Why triangles?

Triangles are a basic block for creating more complex shapes and surfaces





Detailed surface modeled by tiny triangles

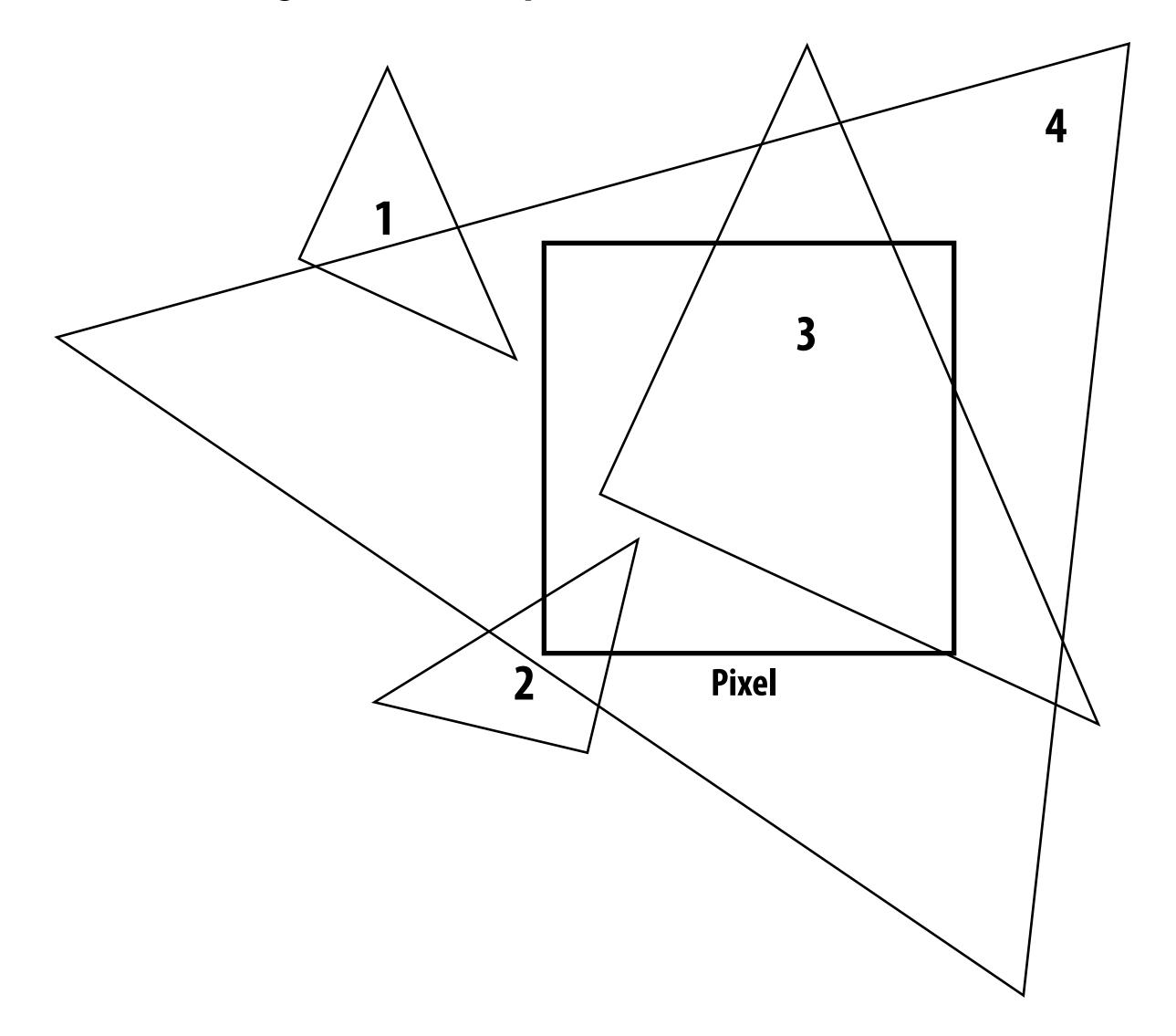
(one pixel)

Triangles - a fundamental primitive

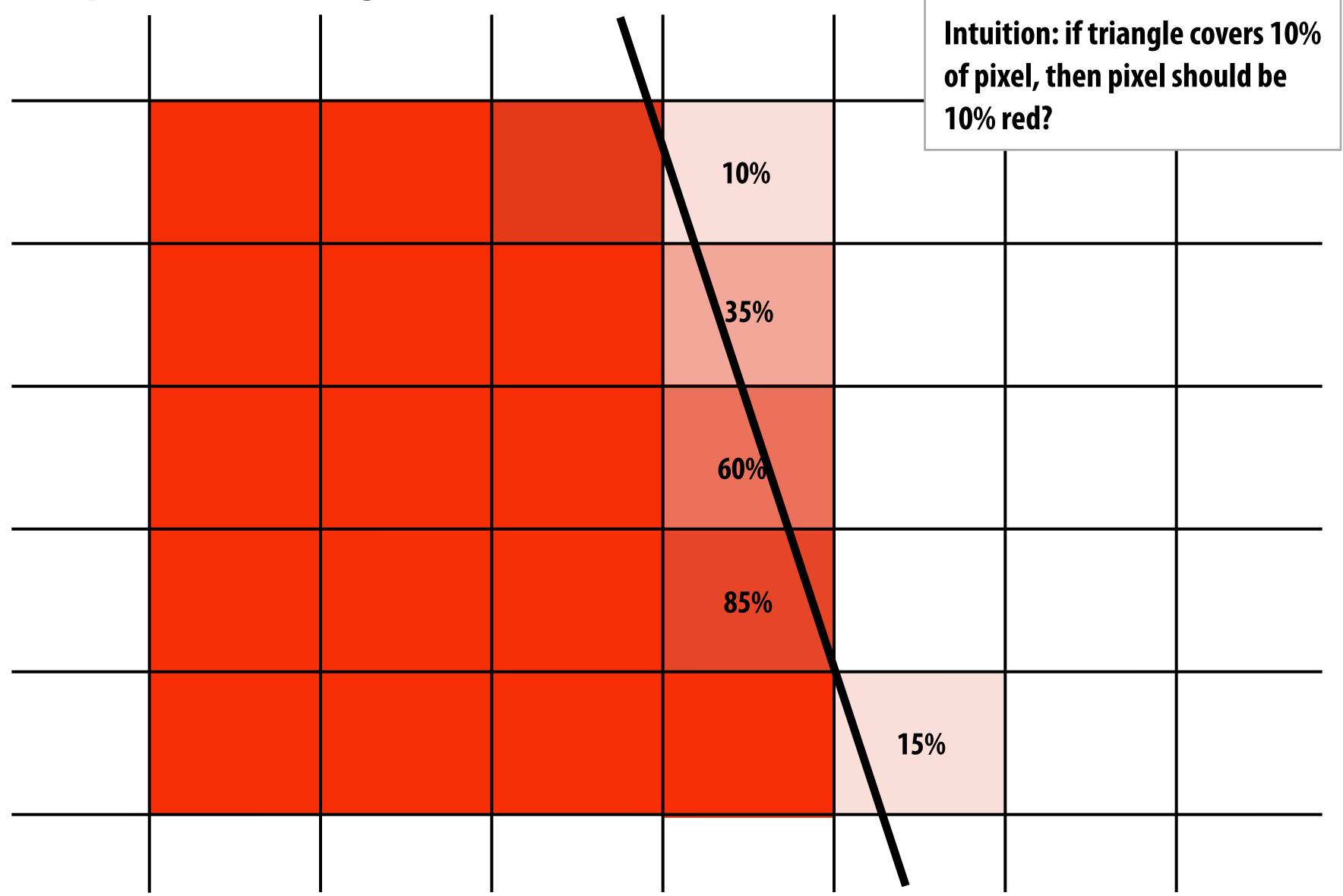
- Why triangles?
 - Most basic polygon
 - Can break up other polygons into triangles
 - Allows programs to optimize one implementation
 - Triangles have unique properties
 - **Guaranteed to be planar**
 - **Well-defined interior**
 - Well-defined method for interpolating values at vertices over triangle (a topic of a future lecture)

What does it mean for a pixel to be covered by a triangle?

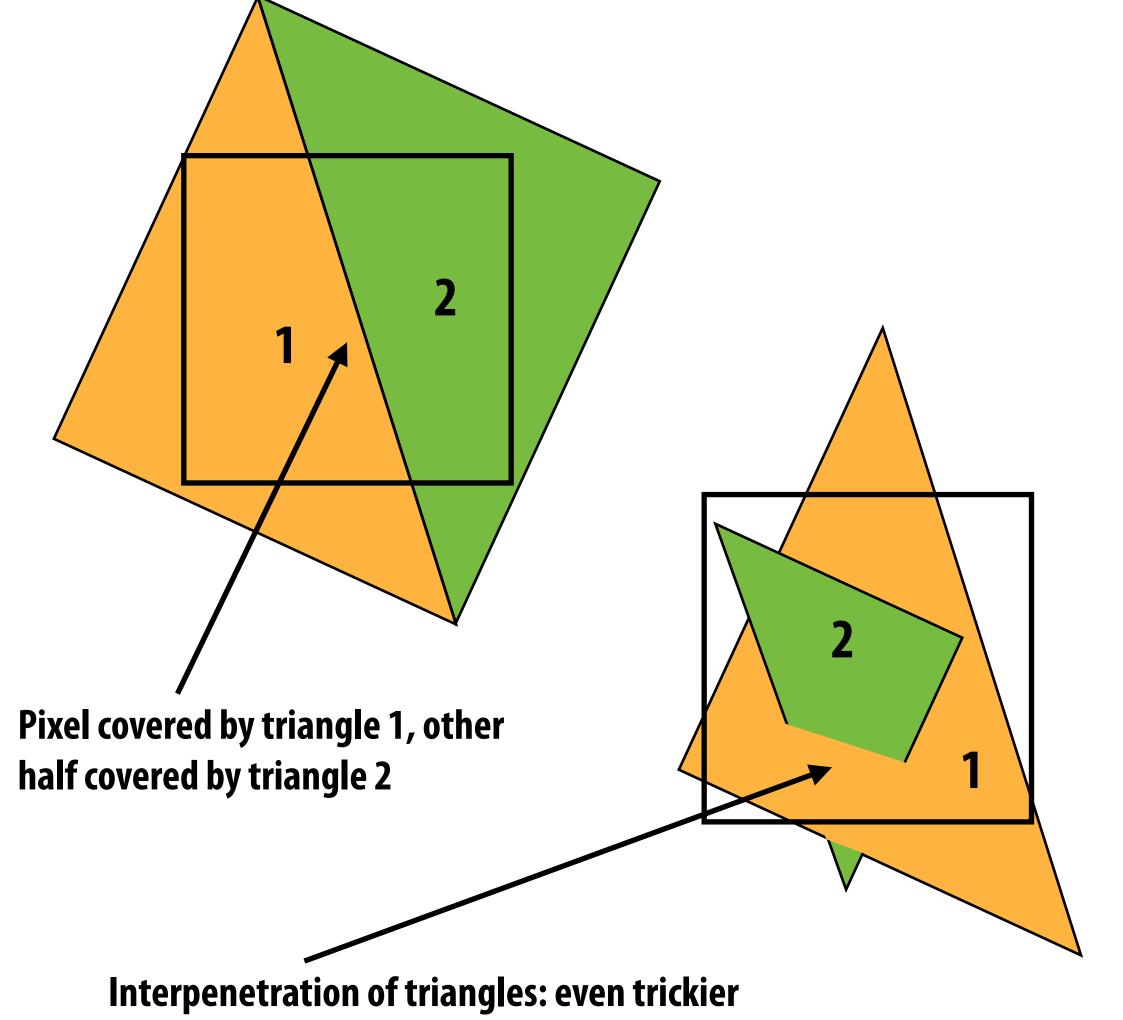
Question: which triangles "cover" this pixel?

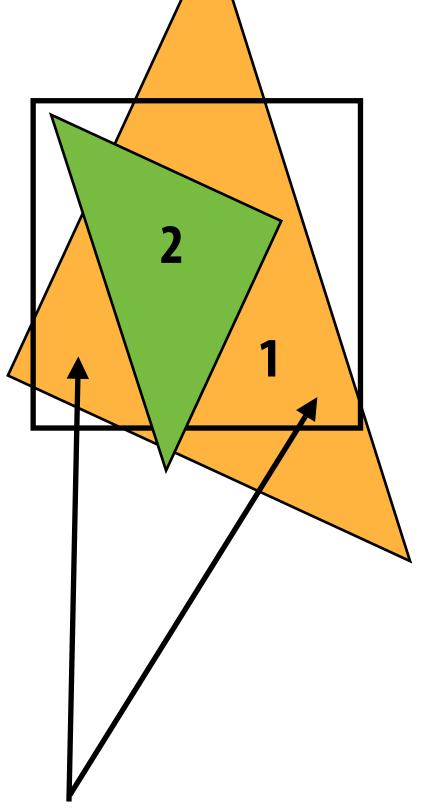


One option: compute fraction of pixel area covered by triangle, then color pixel according to this fraction.



Analytical coverage schemes get tricky when considering occlusion of one triangle by another



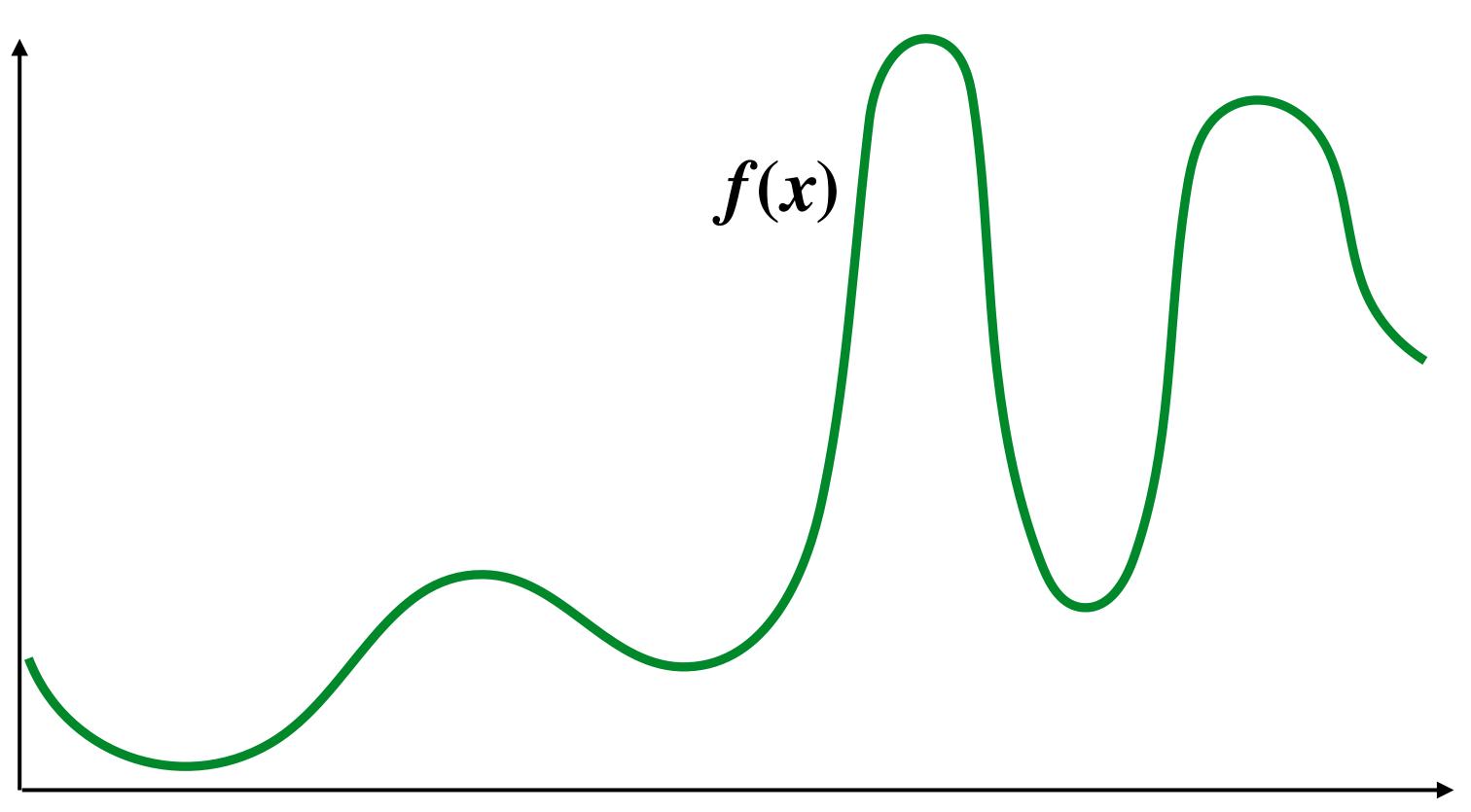


Two regions of triangle 1 contribute to pixel. One of these regions is not even convex.

Today we will draw triangles using a simple method: point sampling

(let's consider sampling in 1D first)

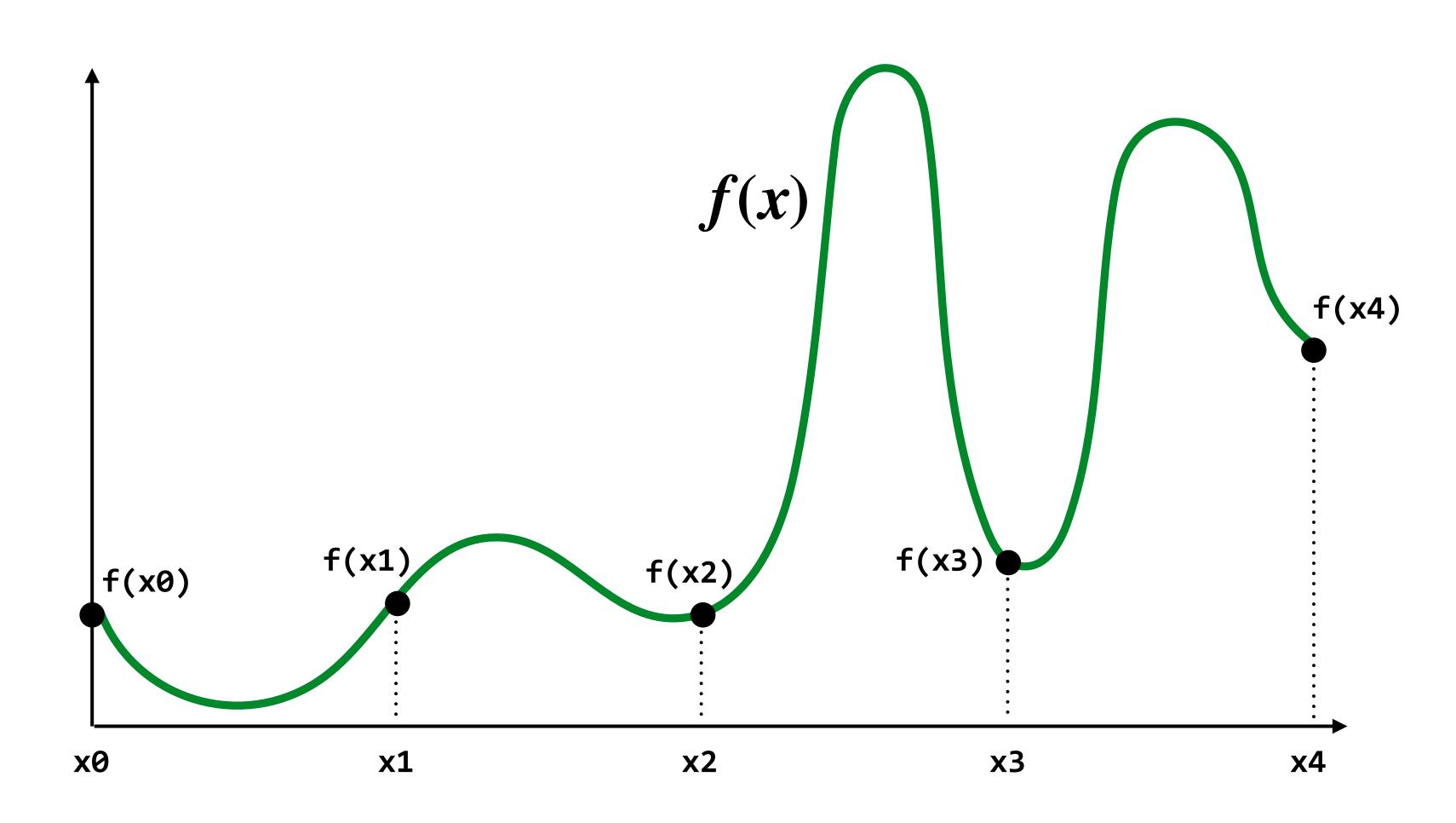
Consider a 1D signal: f(x)



X

Sampling: taking measurements of a signal

Below: five measurements ("samples") of f(x)



Audio file: stores samples of a 1D signal

Audio is often sampled at 44.1 KHz



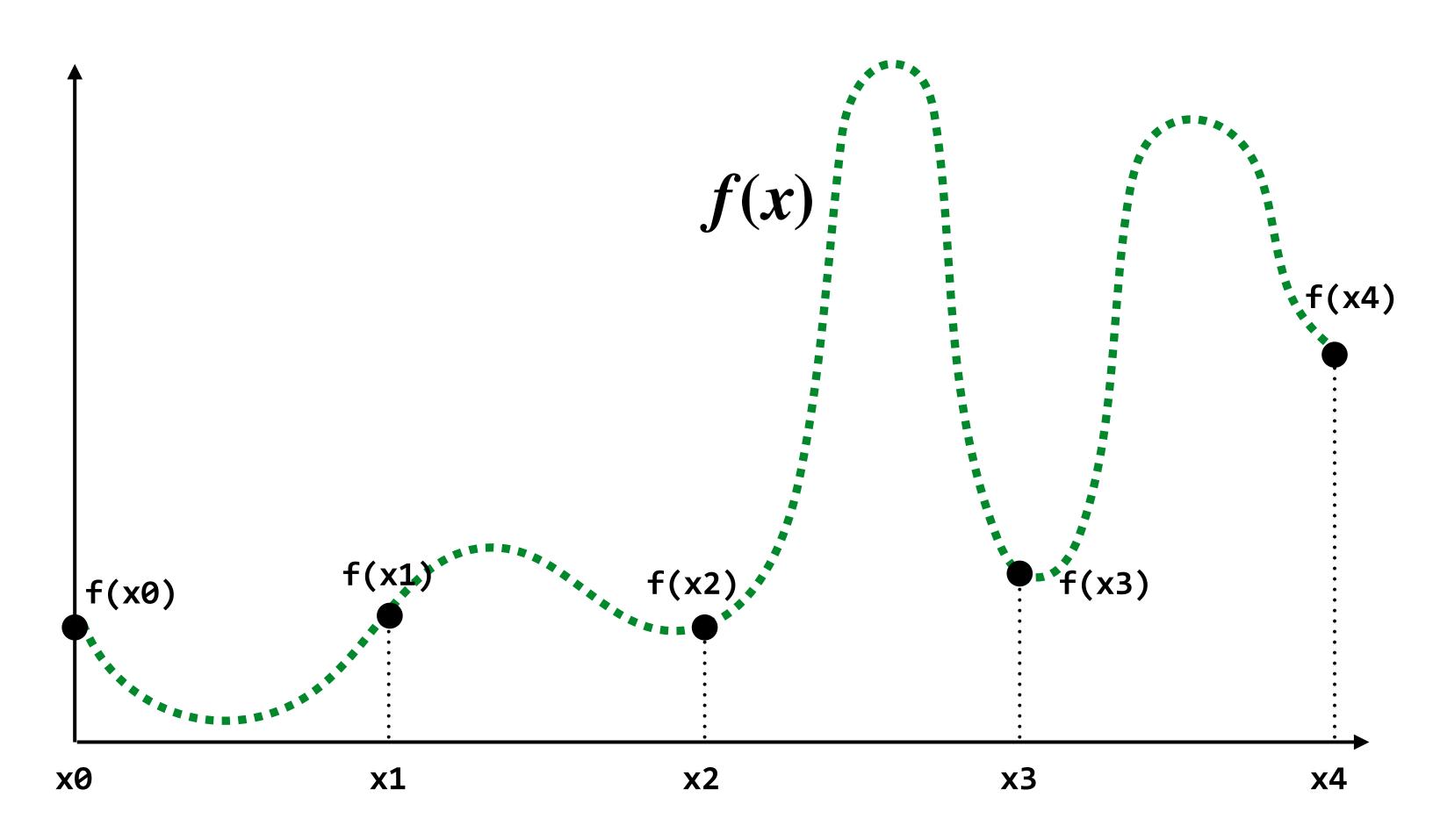
Sampling a function

- Evaluating a function at a point is sampling the function's value
- We can discretize a function by periodic sampling

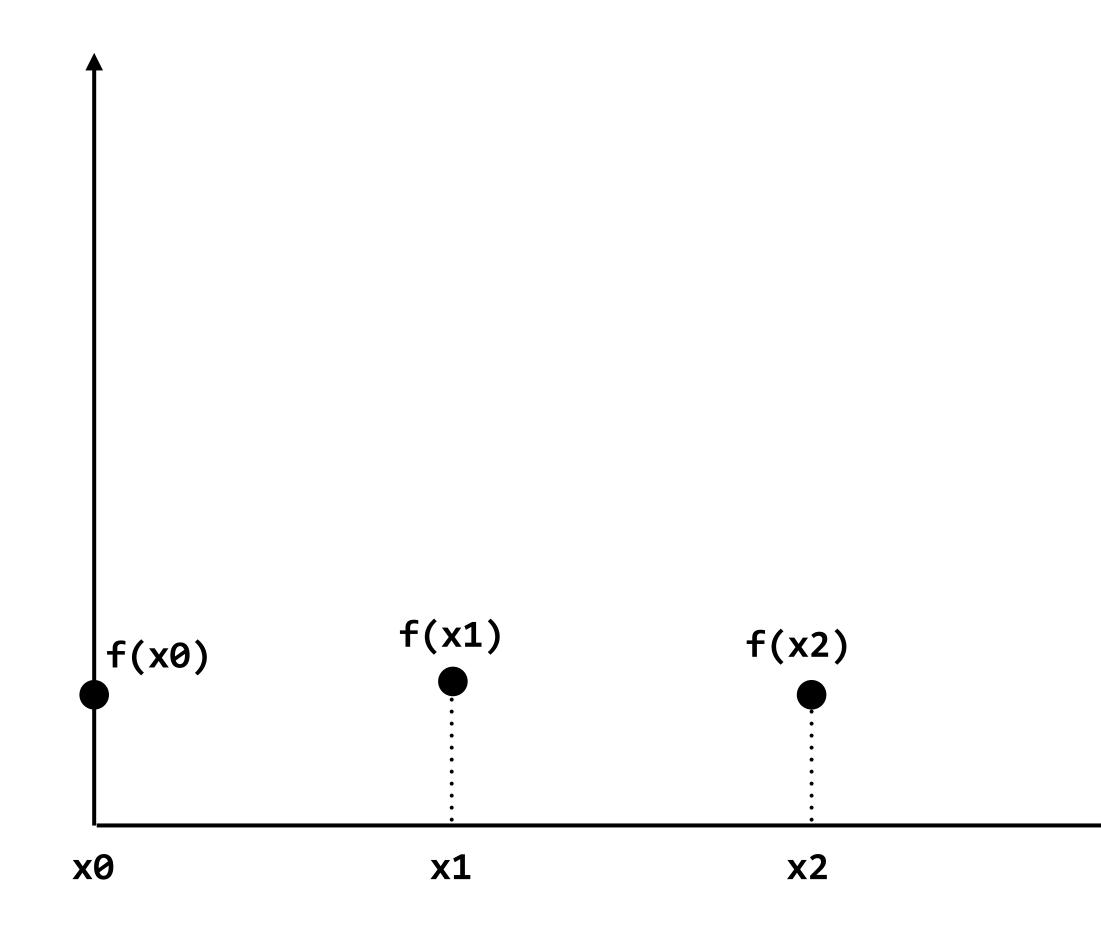
for(int x = 0; x < xmax; x++) output[x] = f(x);

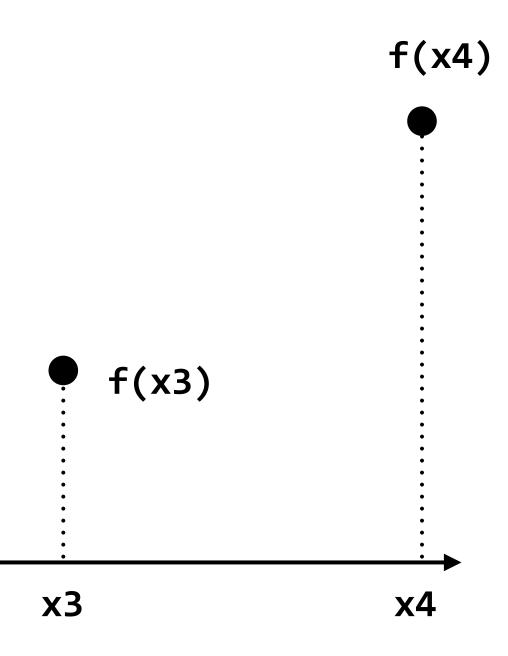
Sampling is a core idea in graphics. In this class we'll sample time (1D), area (2D), angle (2D), volume (3D), etc ...

Reconstruction: given a set of samples, how might we attempt to reconstruct the original signal f(x)?



Reconstruction: given a set of samples, how might we attempt to reconstruct the original signal f(x)?

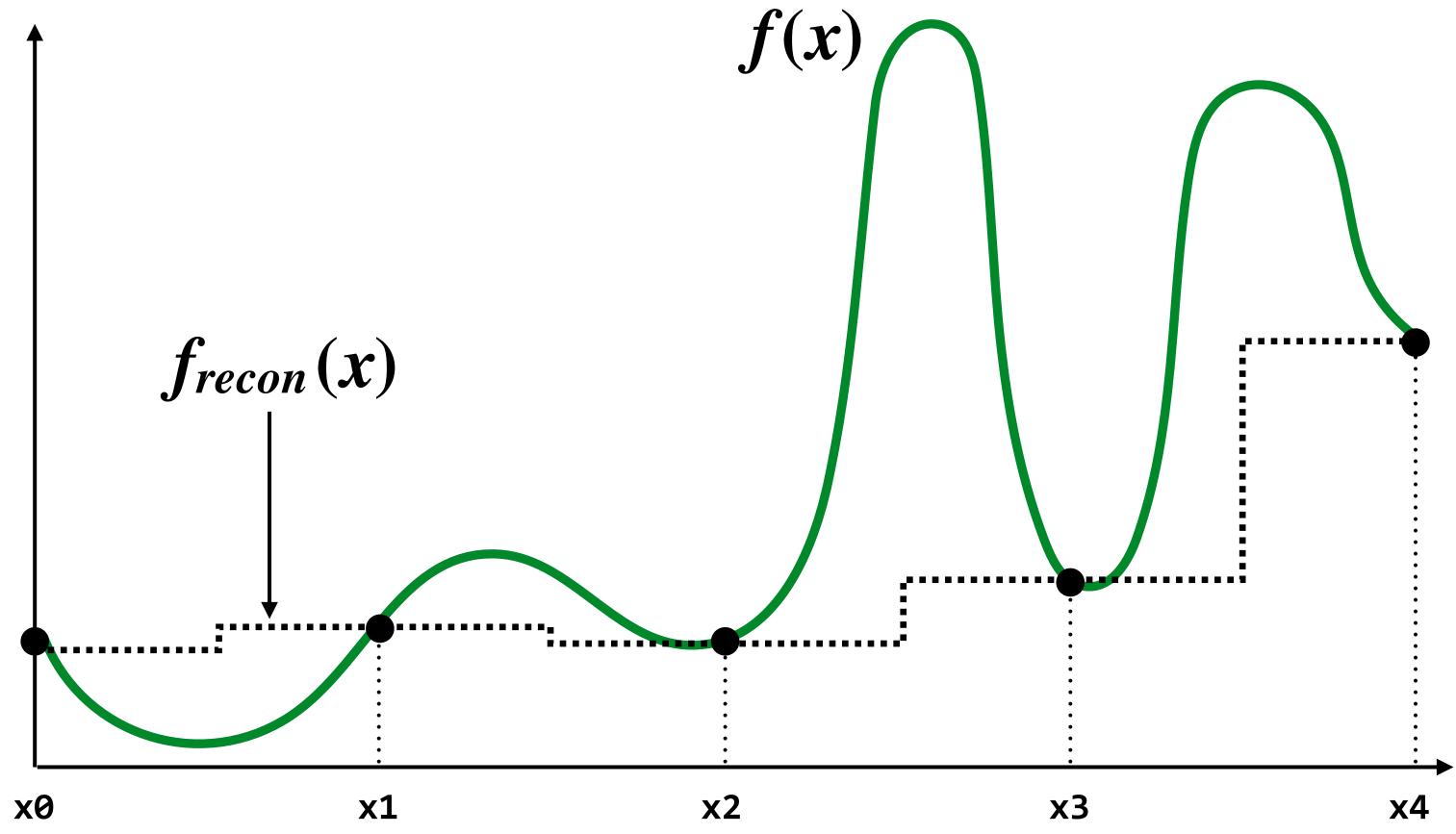




Piecewise constant approximation

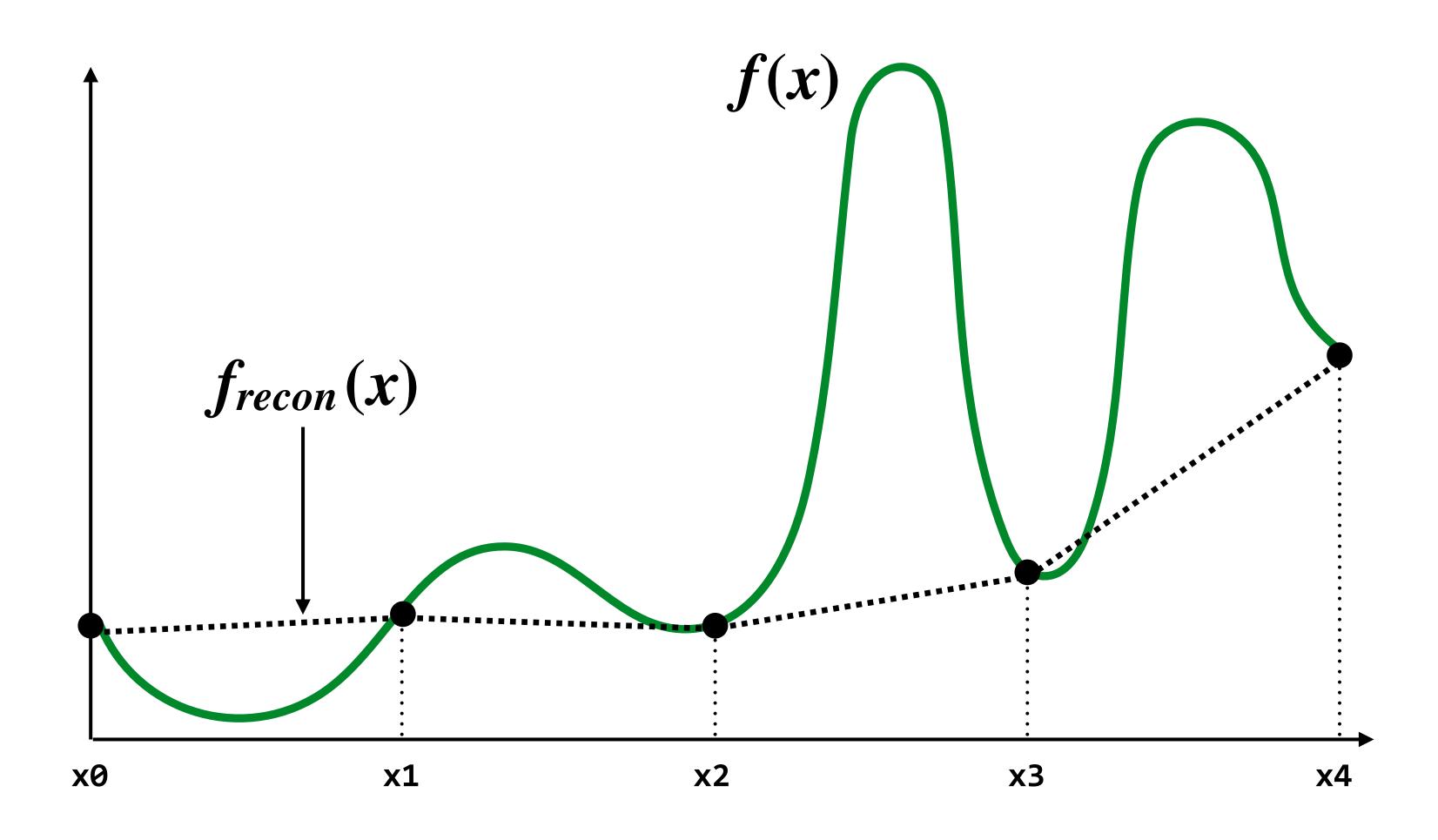
$f_{recon}(x) =$ value of sample closest to x



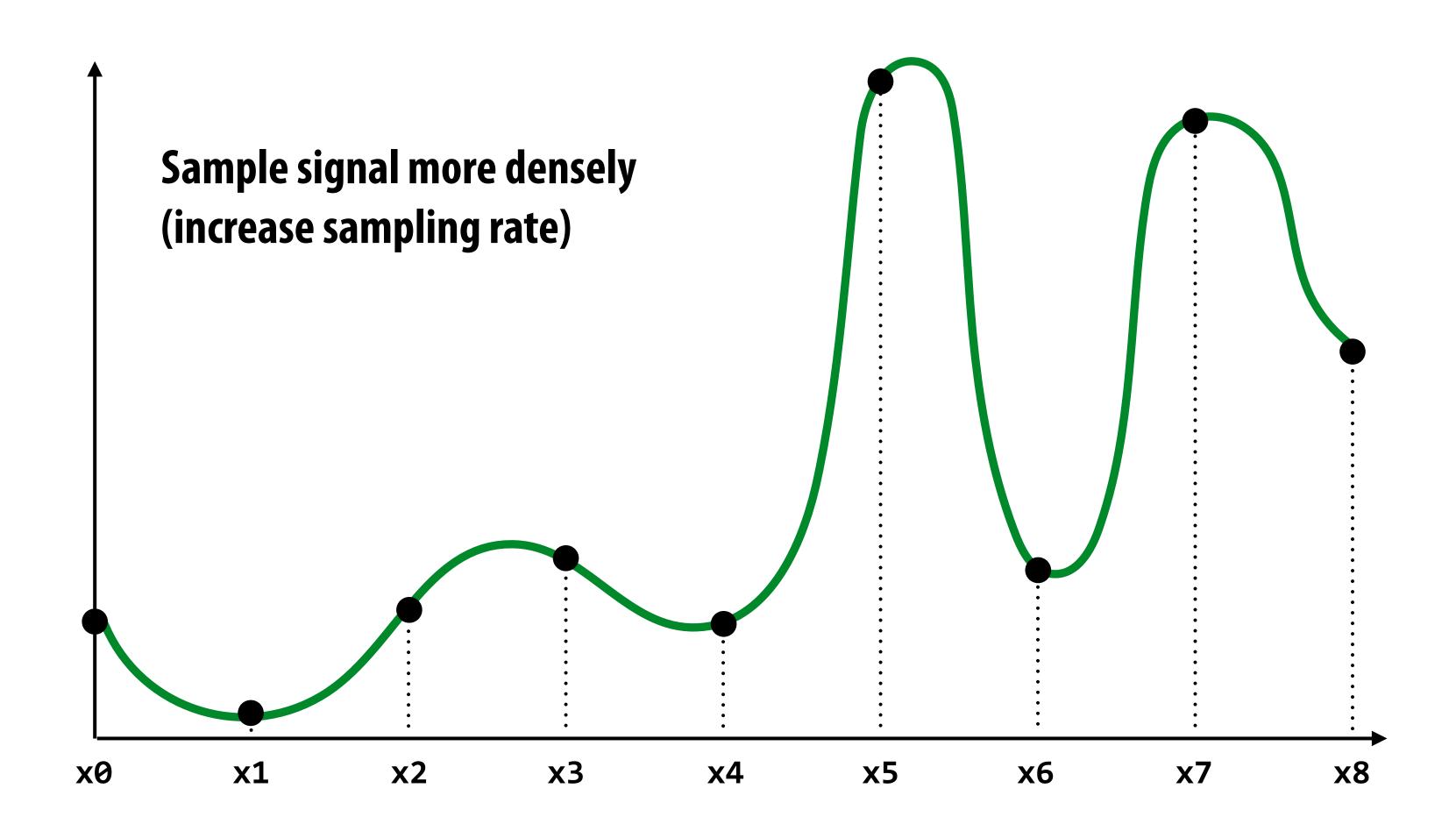


Piecewise linear approximation

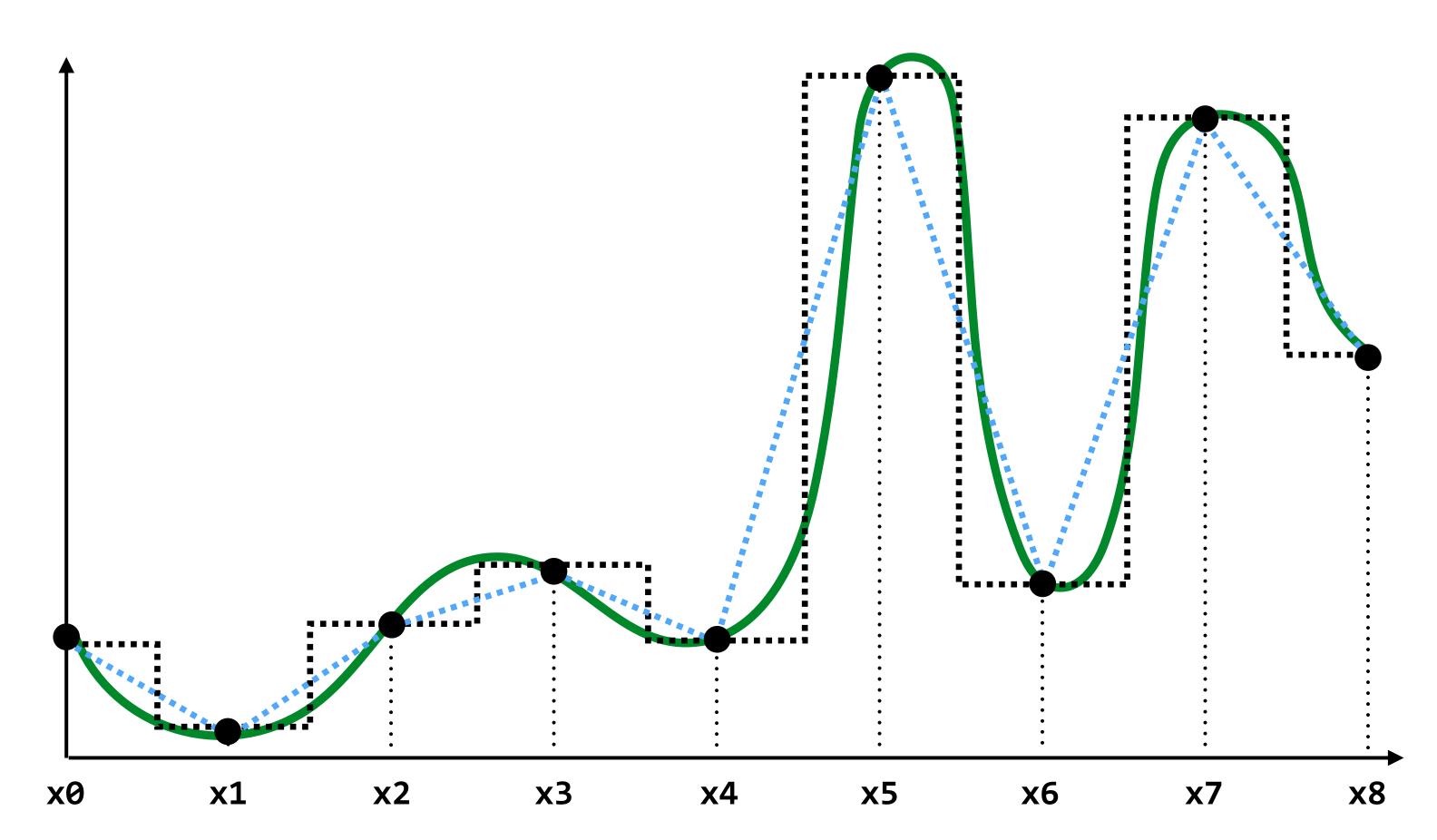
 $f_{recon}(x) =$ linear interpolation between values of two closest samples to x



How can we represent the signal more accurately?



More accurate reconstructions result from denser sampling



••••• = reconstruction via nearest neighbor ••••• = reconstruction via linear interpolation

Drawing a triangle by 2D sampling

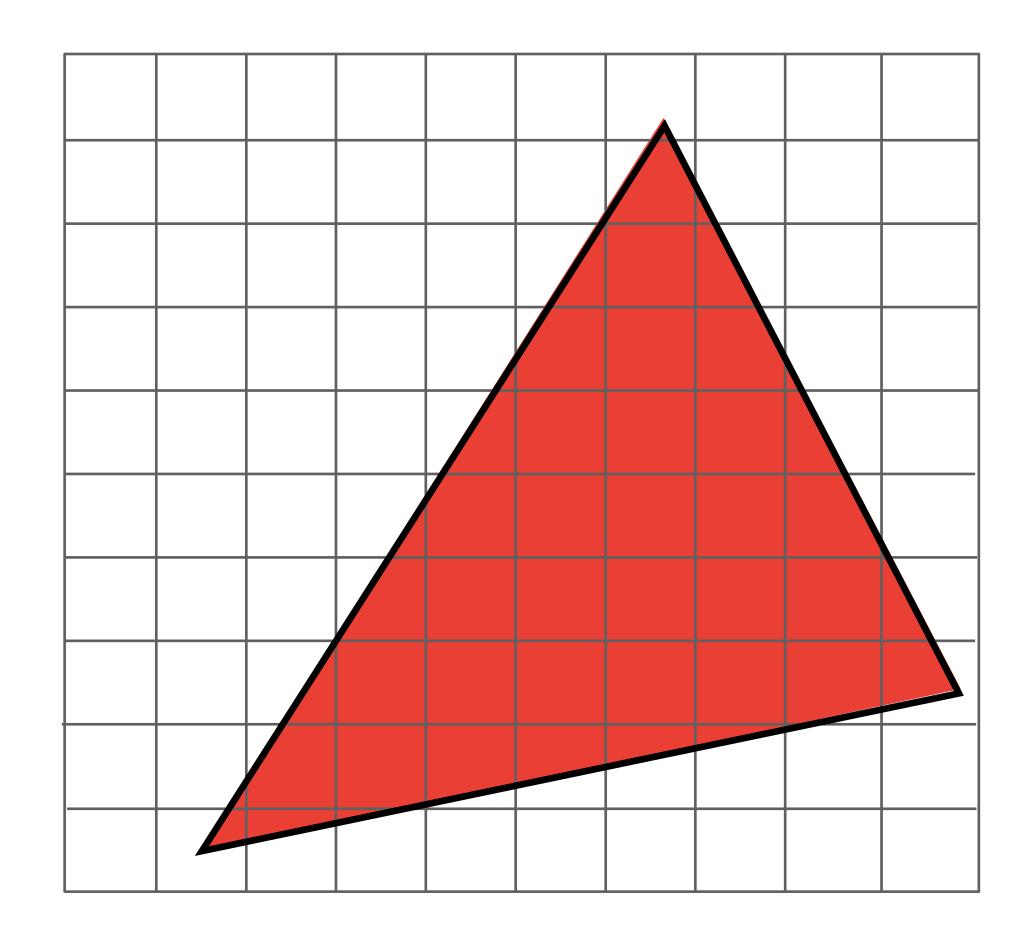
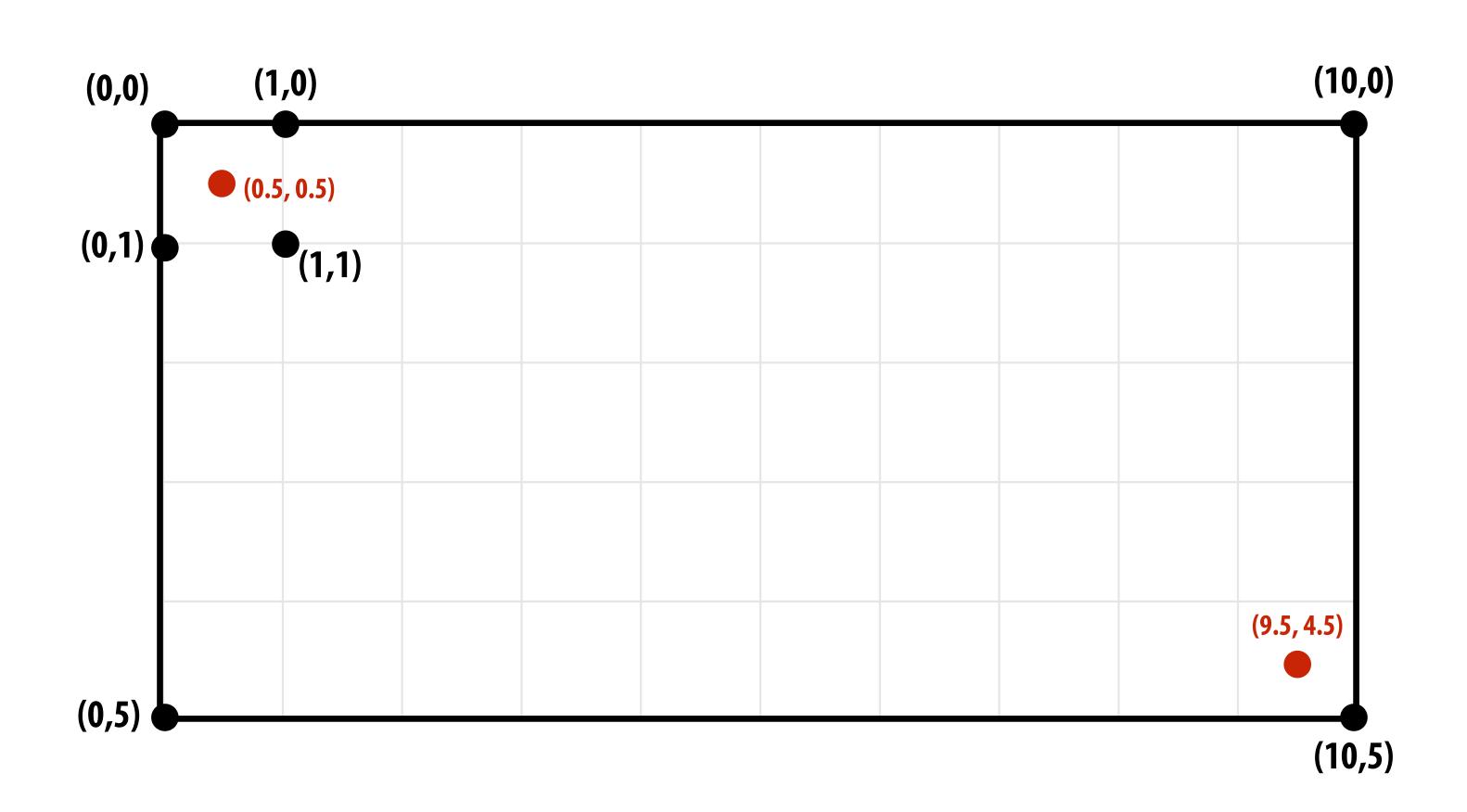


Image as a 2D matrix of pixels

Here I'm showing a 10 x 5 image Identify pixel by its integer (x,y) coordinates

(0,0)	(1,0)				(9,0)
(0,1)	(1,1)				
(0,4)					(9,4)

Continuous coordinate space over image



Define binary function: inside(tri,x,y)

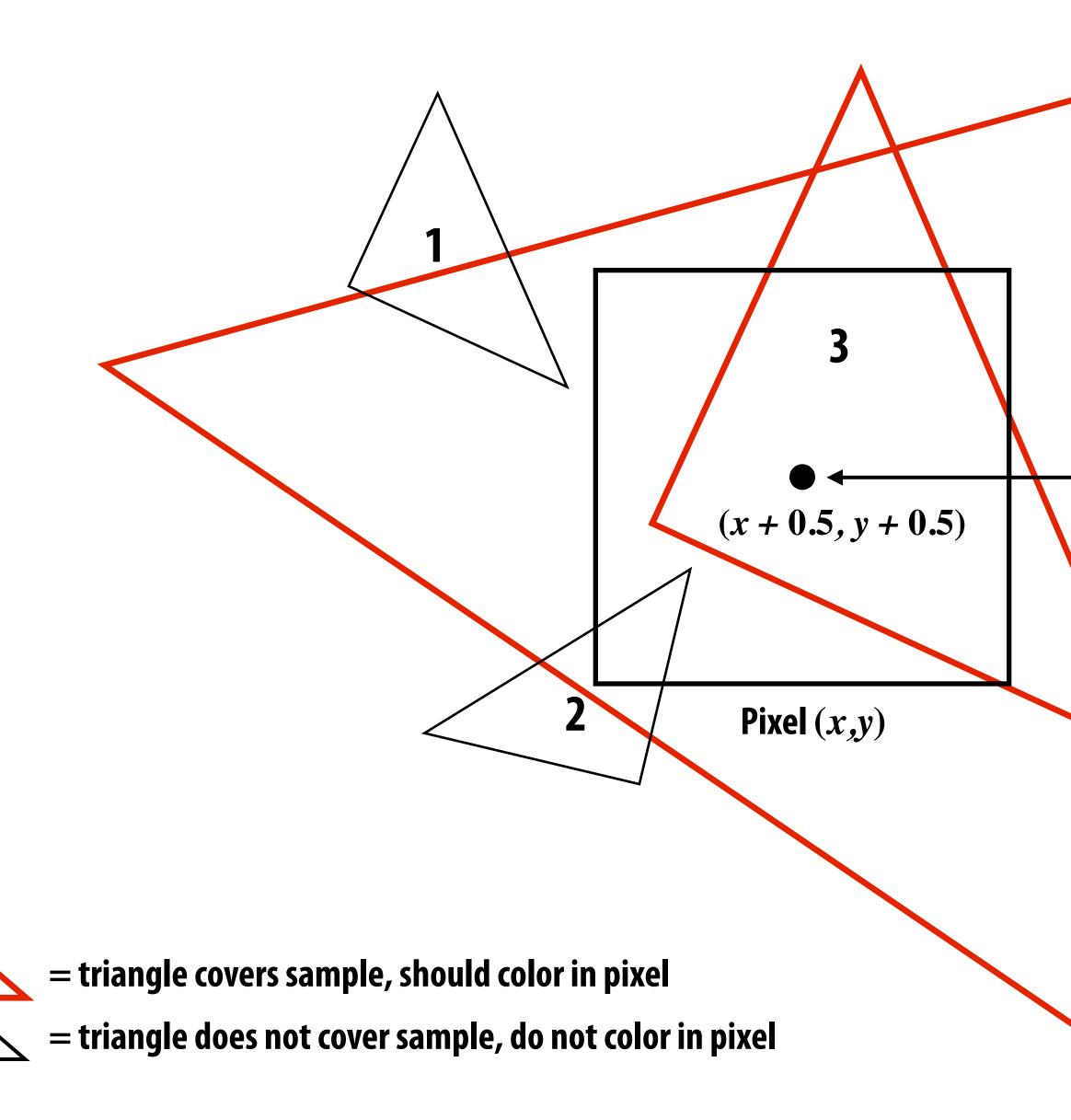
inside(t,x,y) = $\begin{cases} 1 & (x,y) \\ 0 & \text{othe:} \end{cases}$

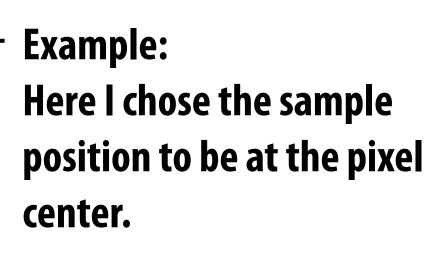
(x,y) in triangle t

otherwise

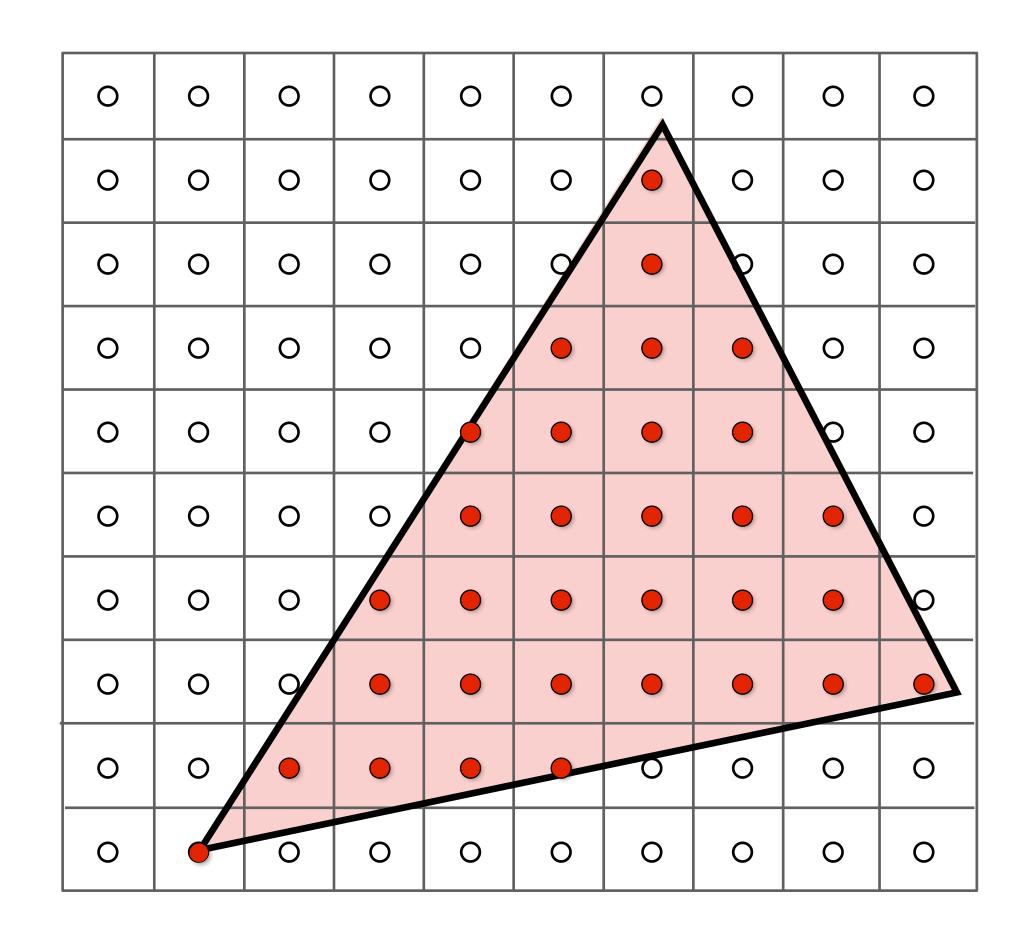
Sampling the binary function: inside(tri,x,y)

4

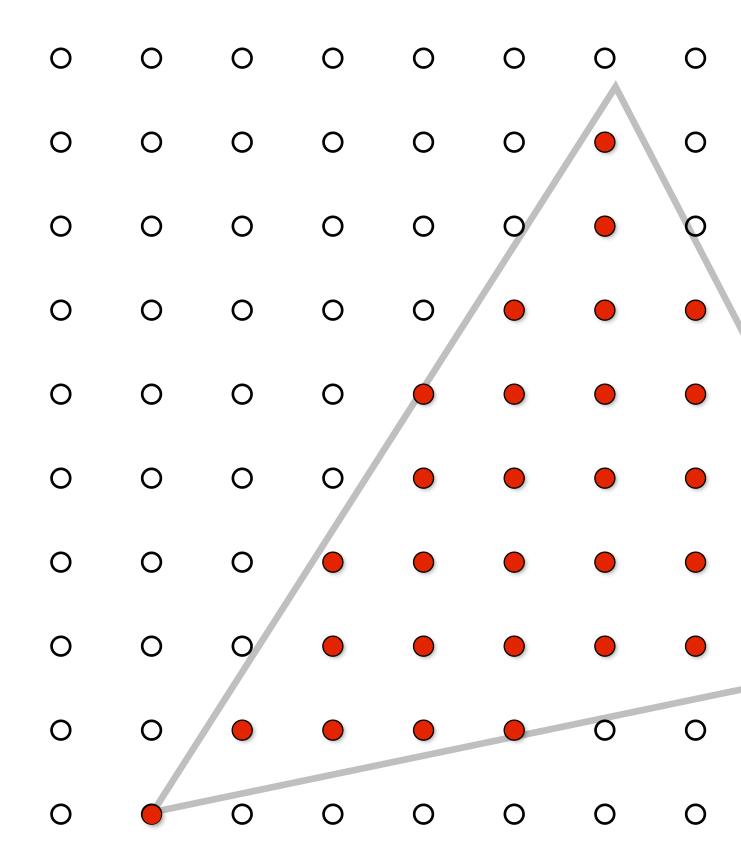


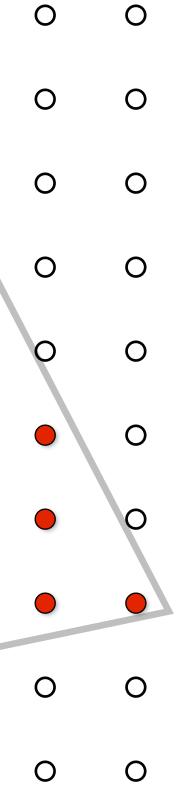


Sample coverage at pixel centers



Sample coverage at pixel centers





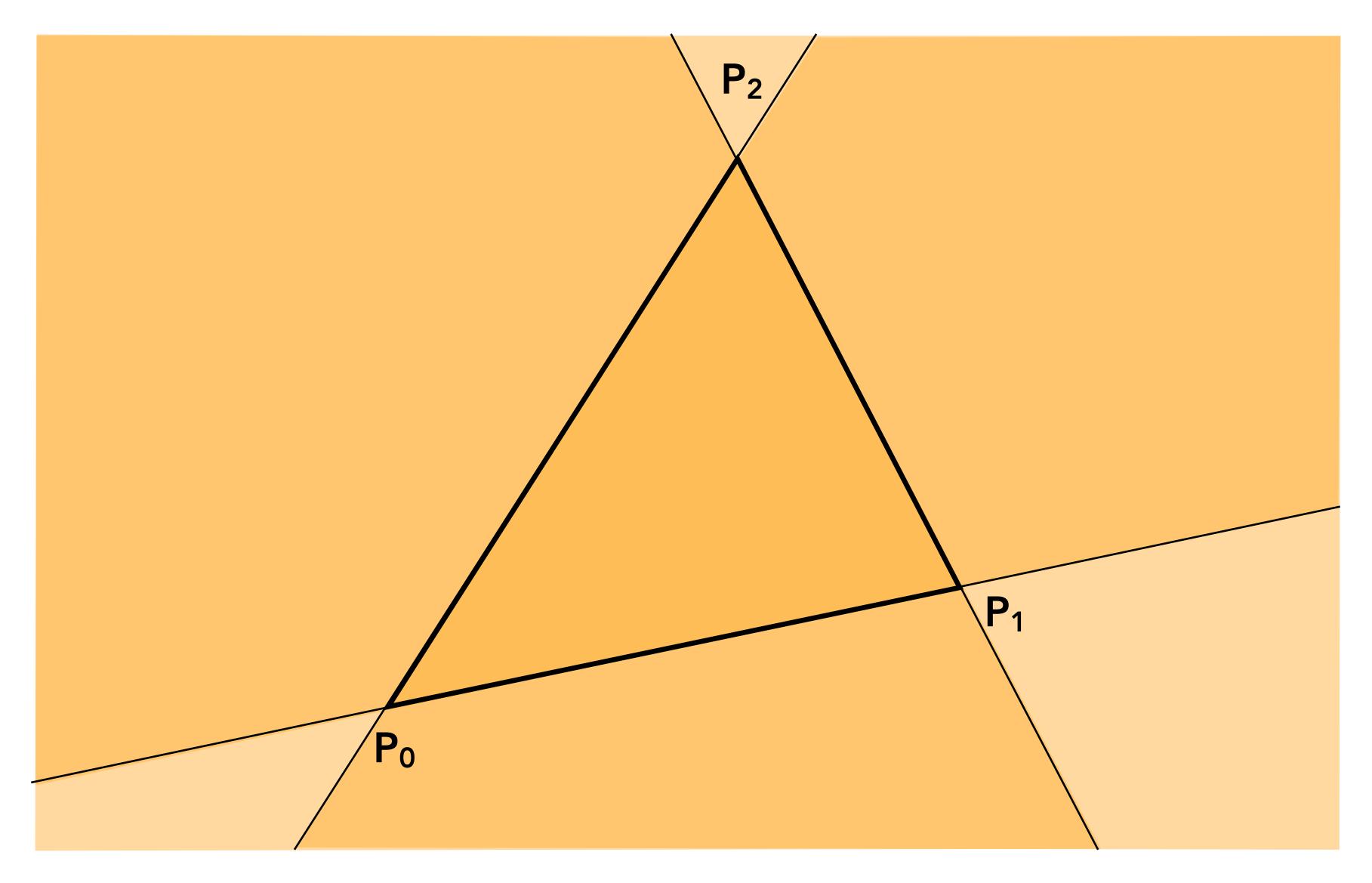
Rasterization = sampling a 2D binary function

for (int x = 0; x < xmax; x++)
for (int y = 0; y < ymax; y++)
image[x][y] = f(x + 0.5, y + 0.5);</pre>

Rasterize triangle tri by sampling the function
f(x,y) = inside(tri,x,y)

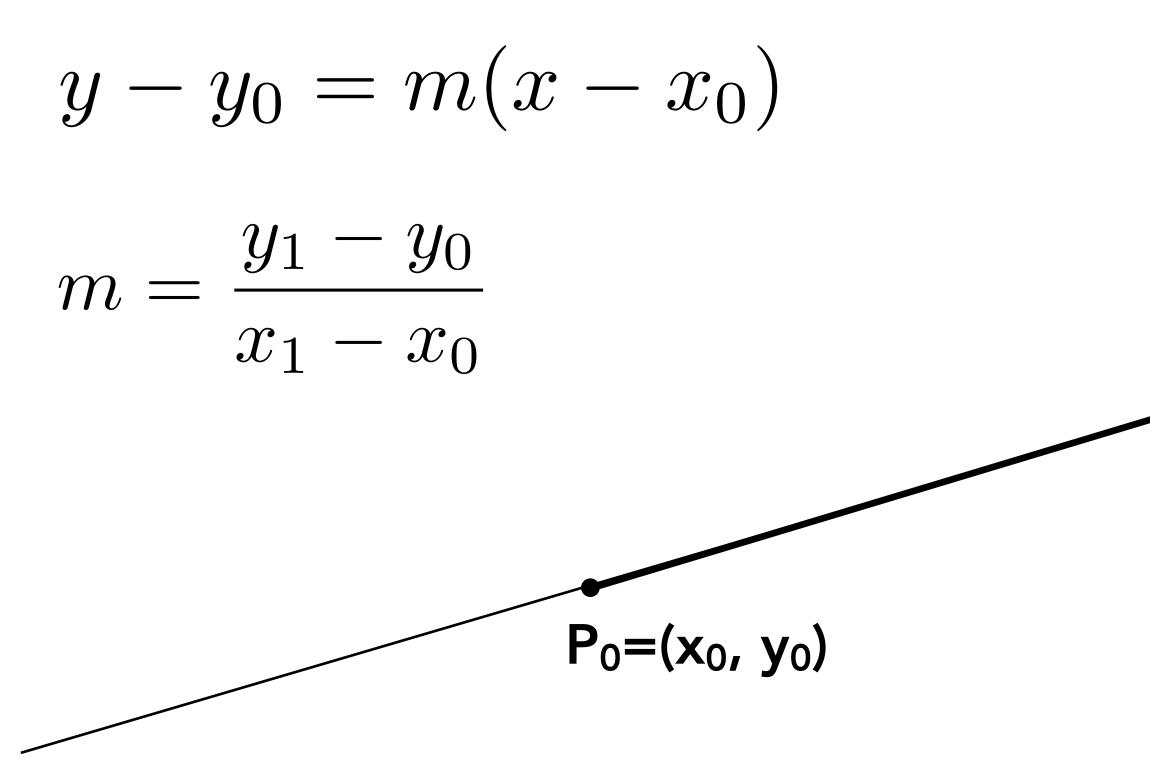
Evaluating inside(tri,x,y)

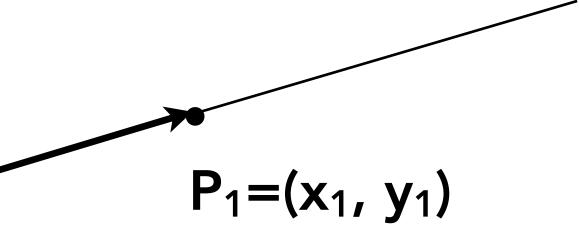
Triangle = intersection of three half planes



Point slope form of a line

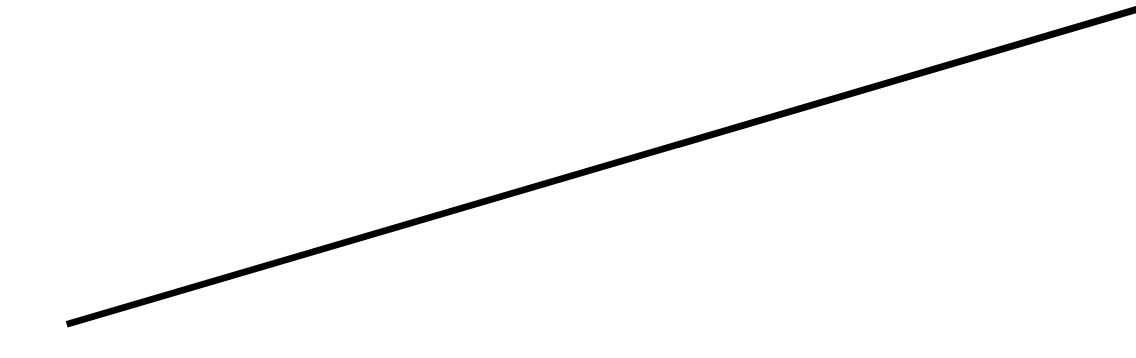
(You might have seen this in high school)

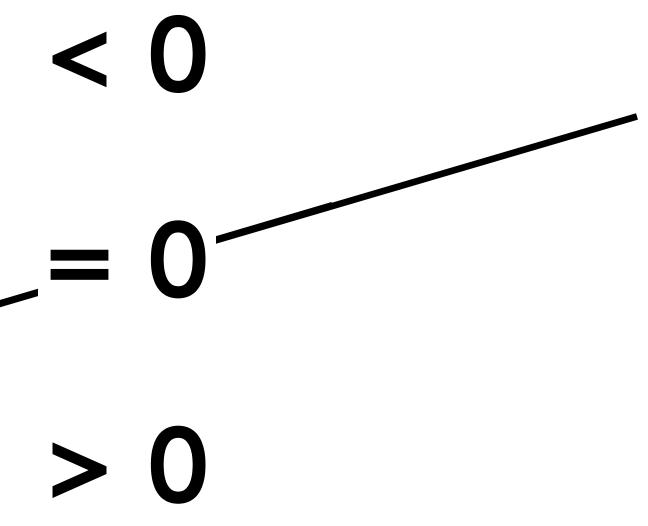


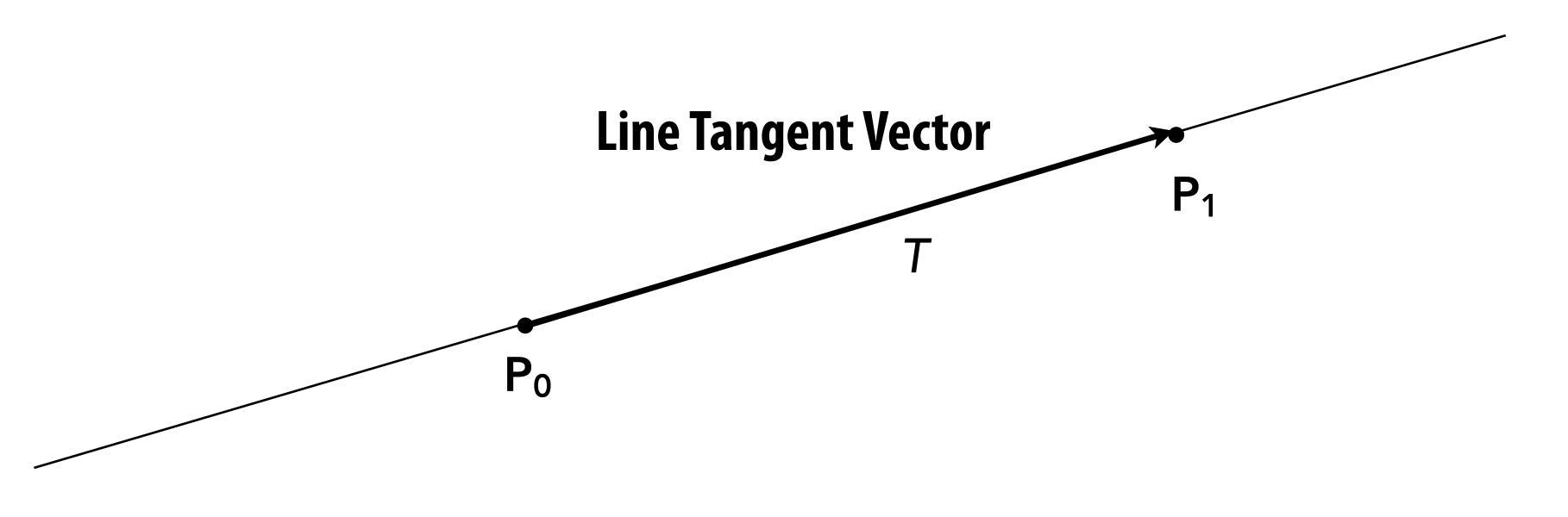


Each line defines two half-planes

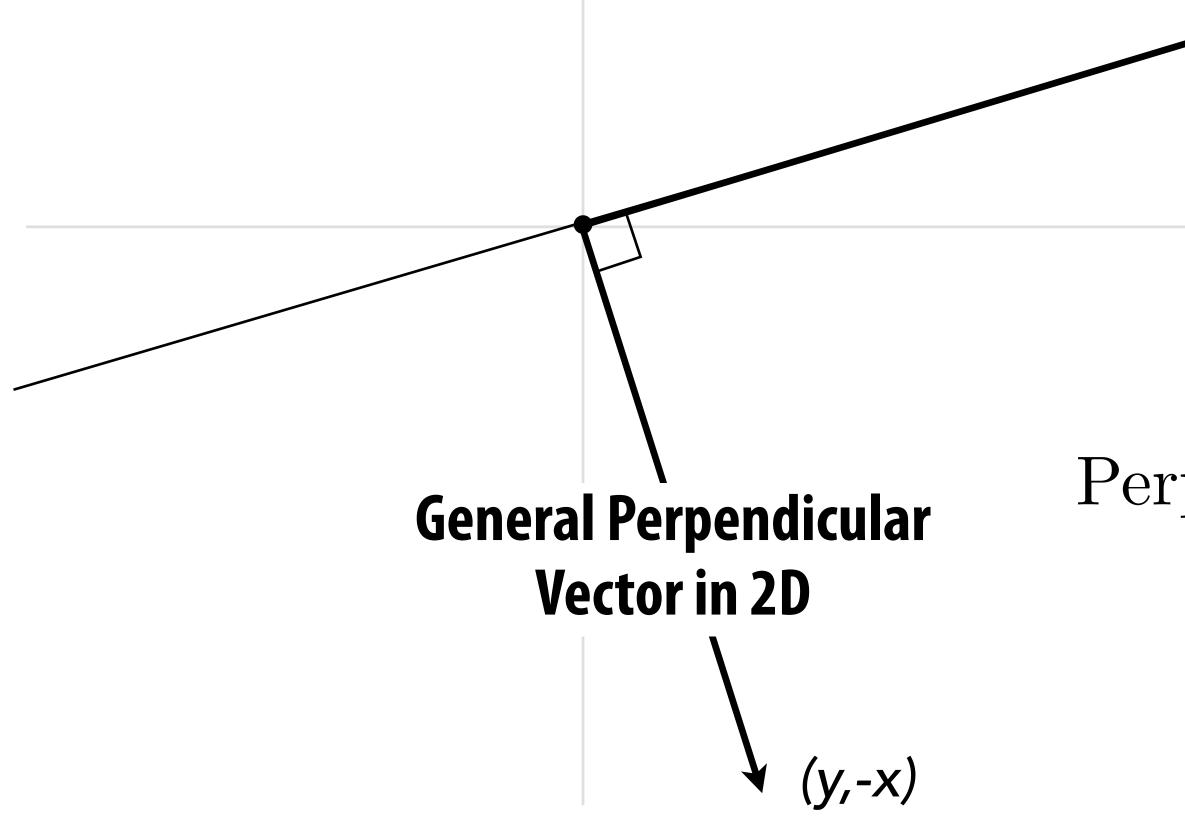
- Implicit line equation
 - L(x,y) = Ax + By + C
 - On the line: L(x,y) = 0
 - "Negative side" of line: L(x,y) < 0
 - "Positive" side of line: L(x,y) > 0

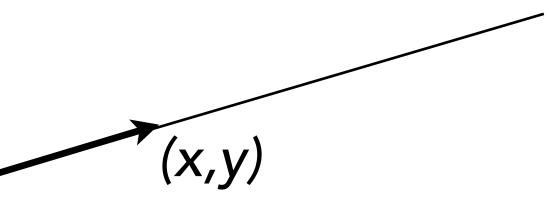






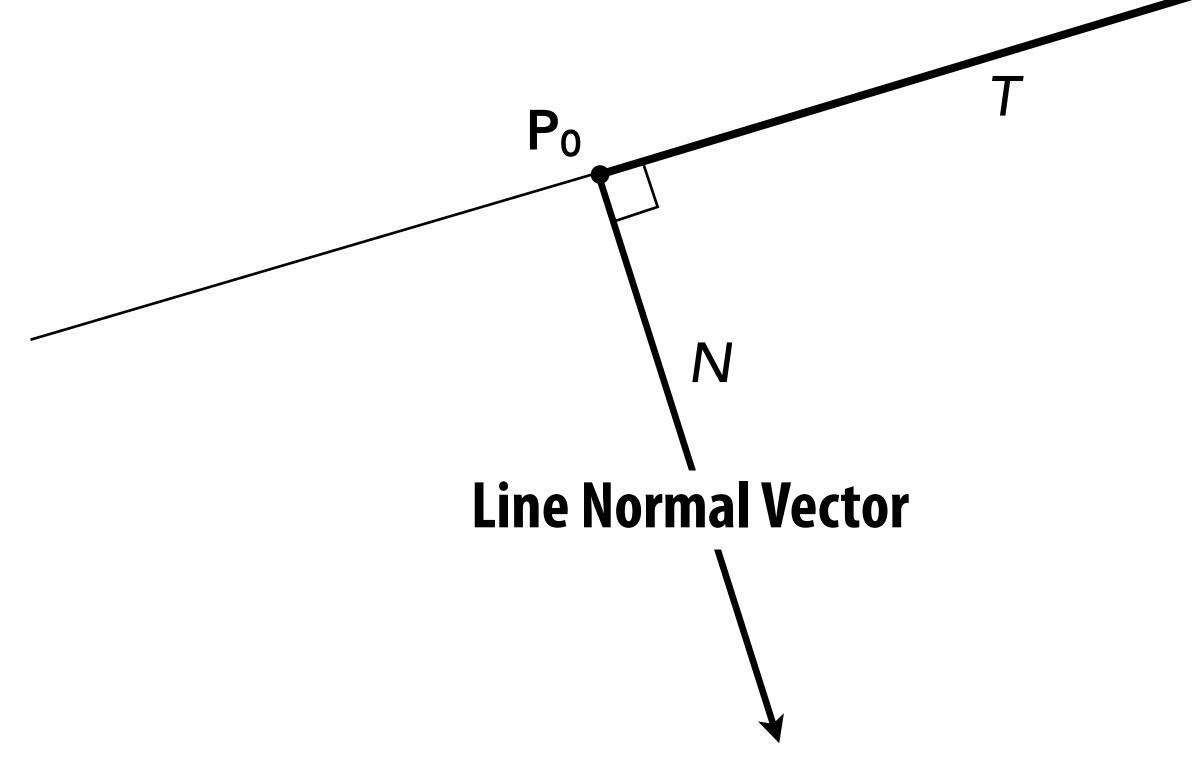
 $T = P_1 - P_0 = (x_1 - x_0, y_1 - y_0)$

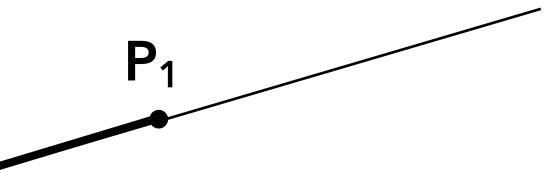




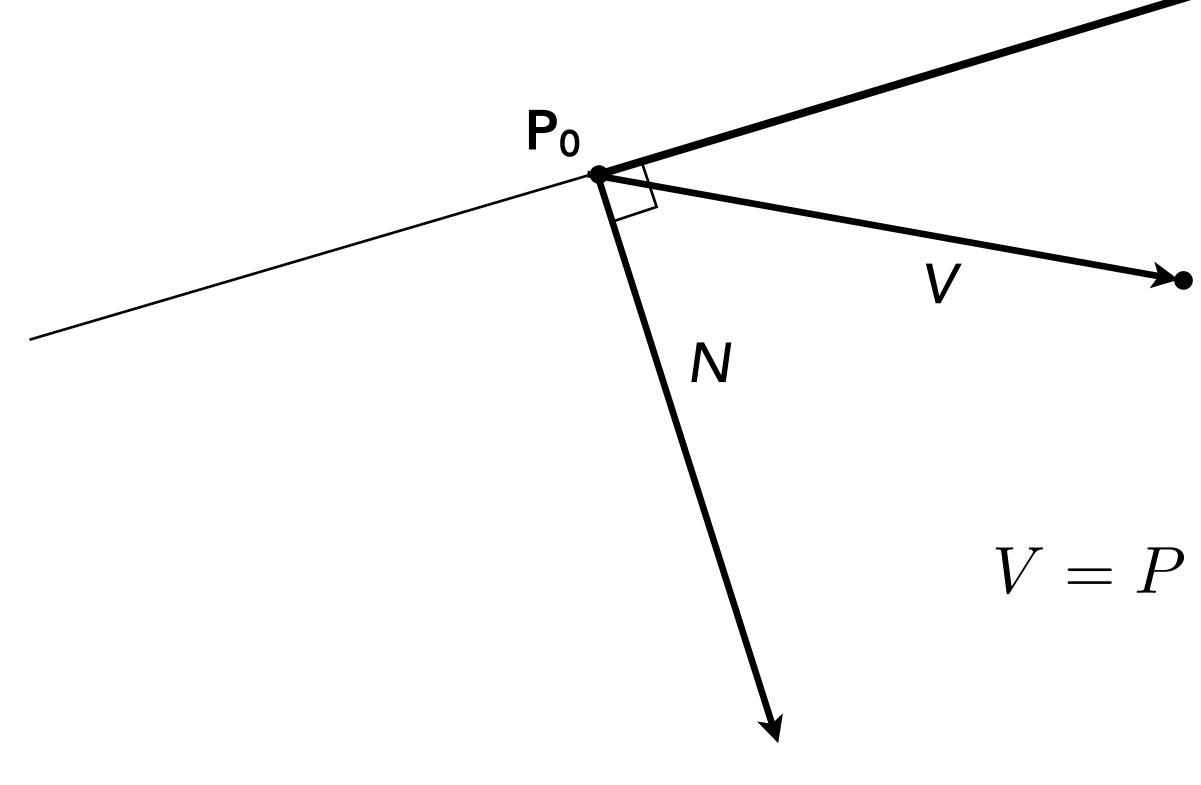
Perp(x, y) = (y, -x)

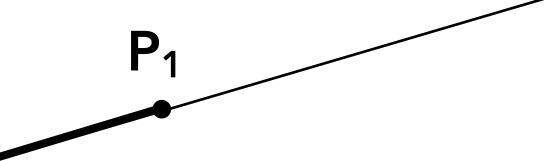
 $N = Perp(T) = (y_1 - y_0, -(x_1 - x_0))$





Now consider a point P=(x,y). Which side of the line is it on?

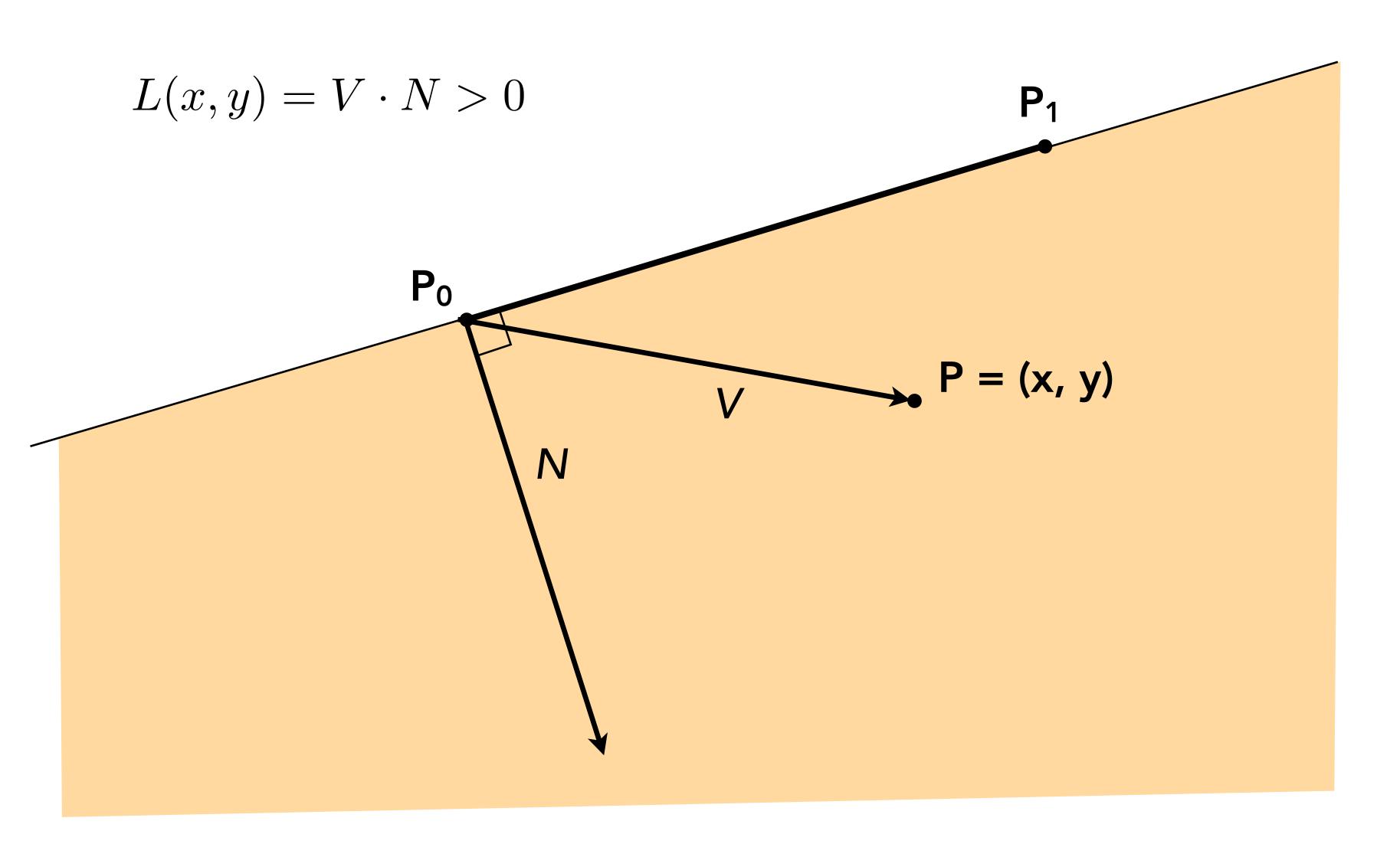




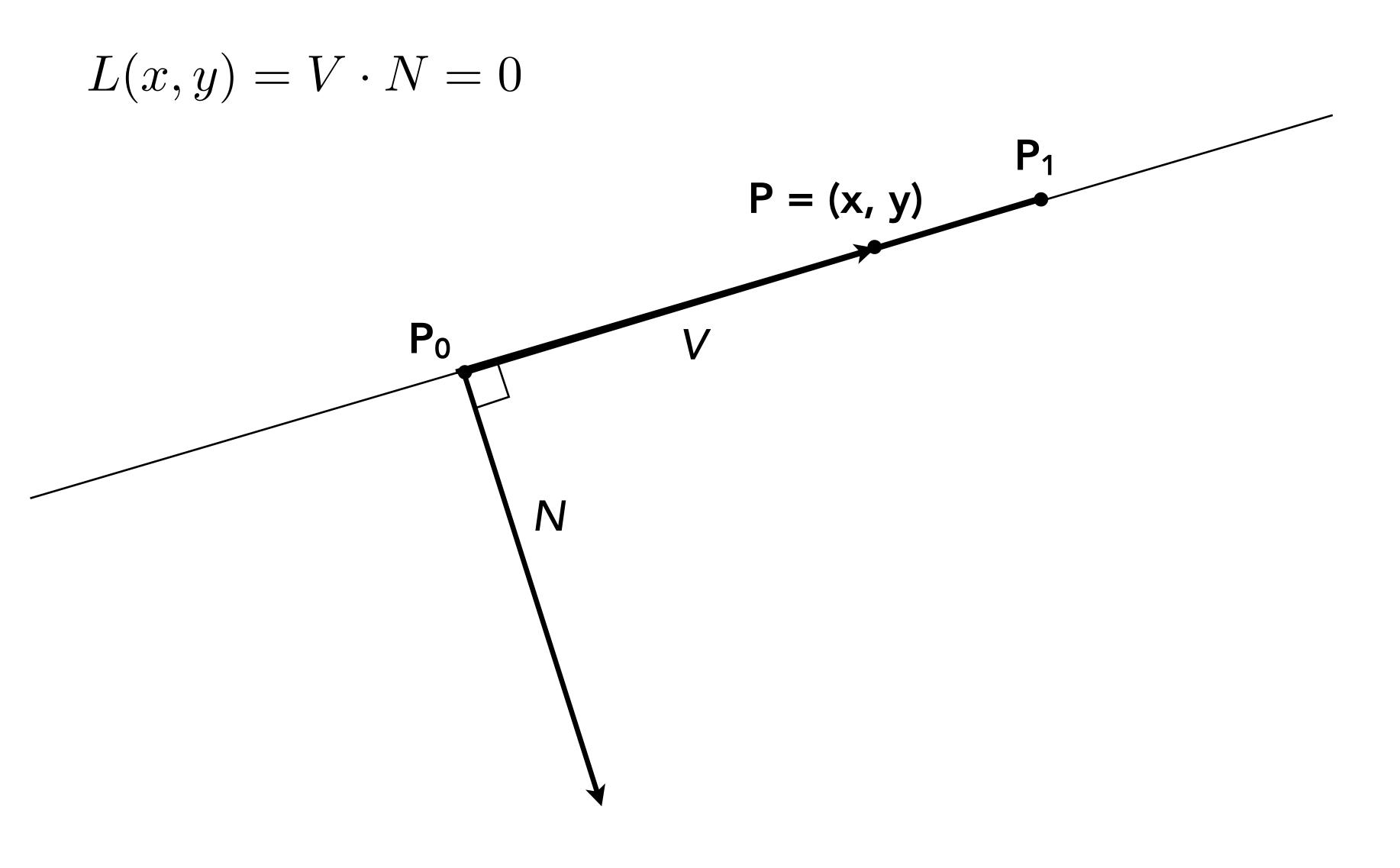
$\mathsf{P}=(\mathsf{x},\,\mathsf{y})$

$V = P - P_0 = (x - x_0, y - y_0)$

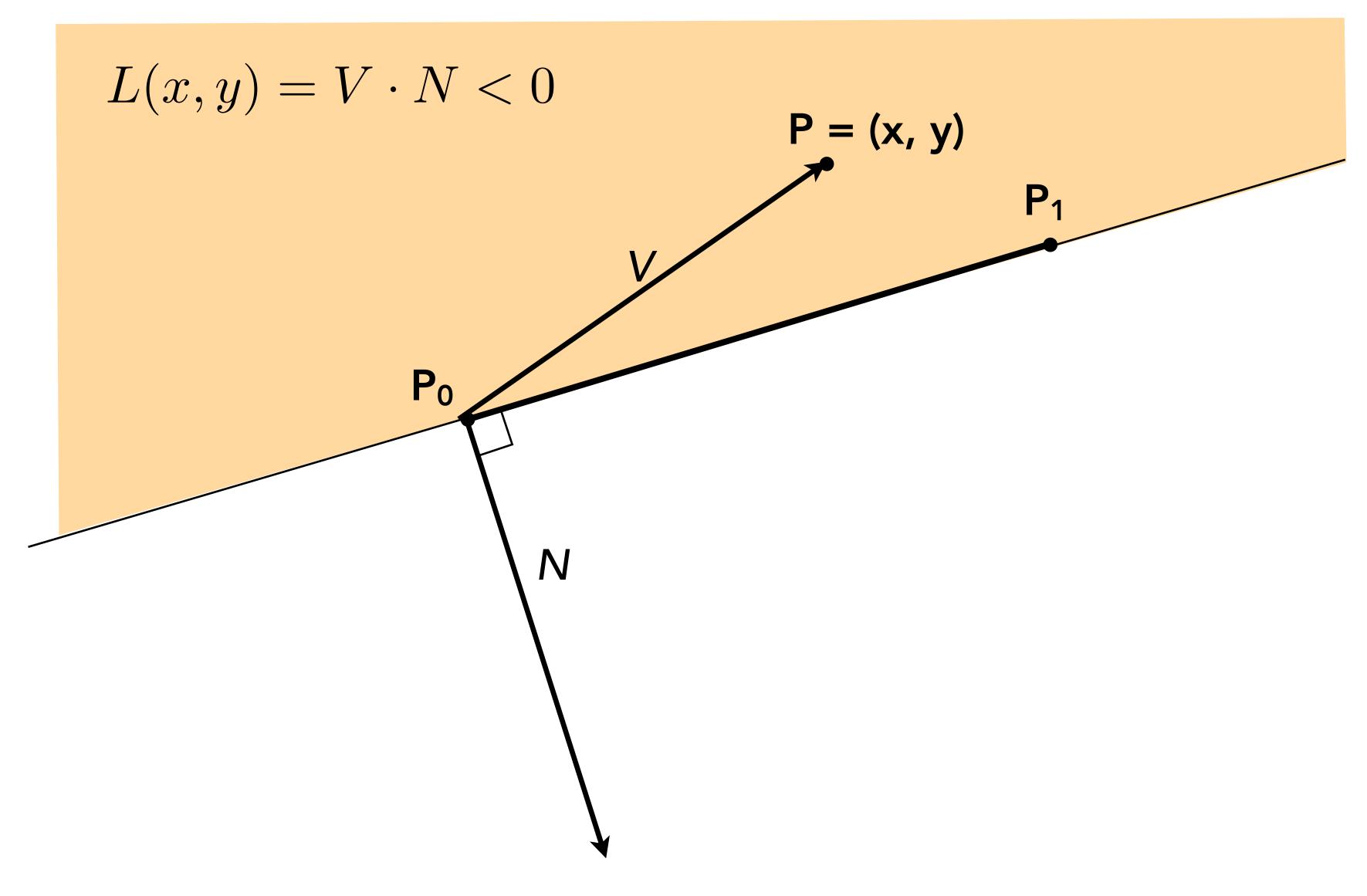
Line equation tests



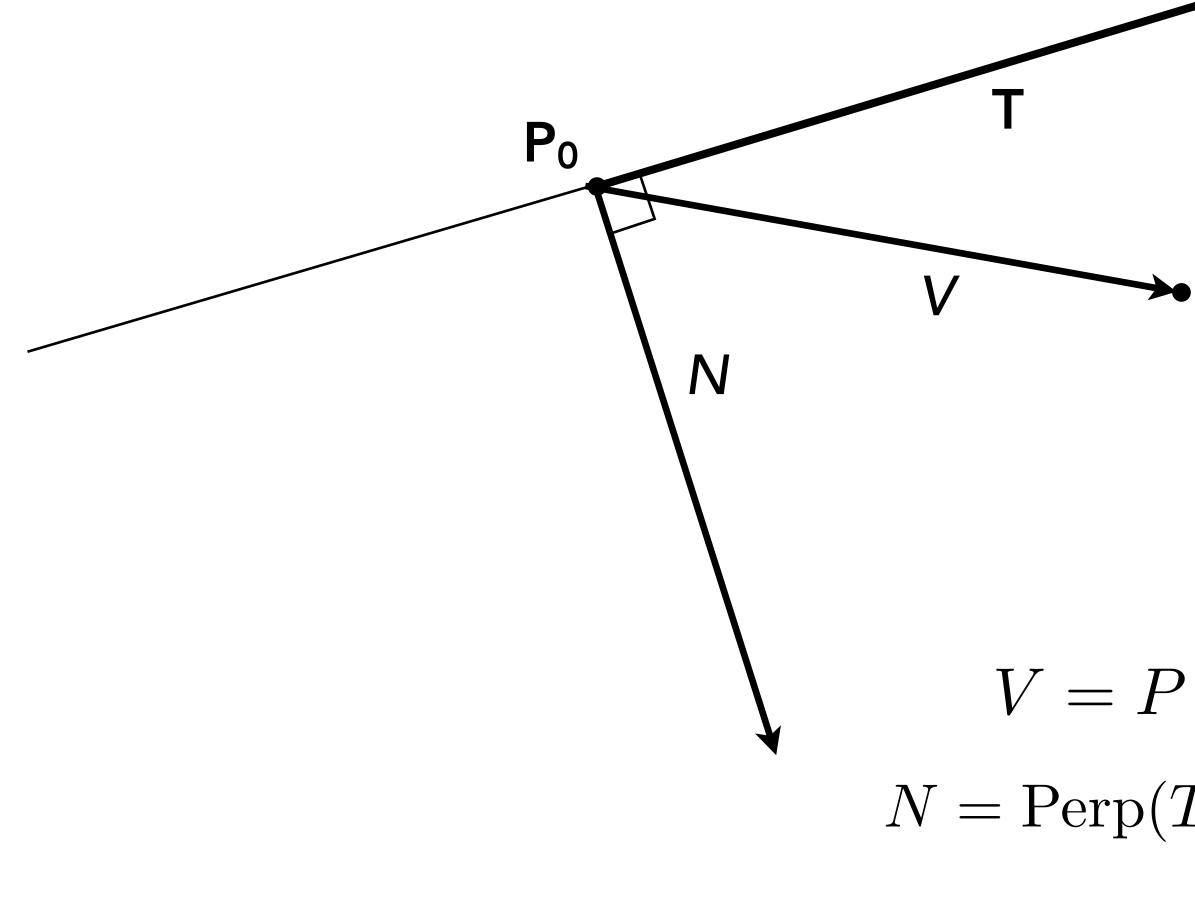
Line equation tests

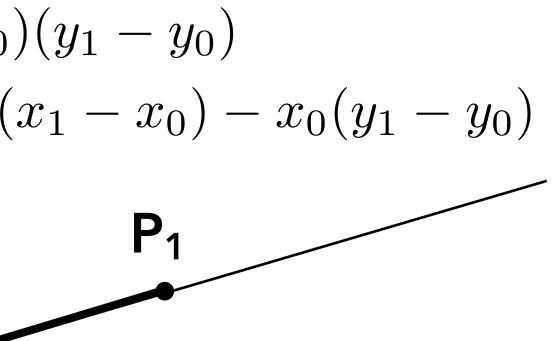


Line equation tests



 $L(x,y) = V \cdot N = -(y - y_0)(x_1 - x_0) + (x - x_0)(y_1 - y_0)$ $= (y_1 - y_0)x - (x_1 - x_0)y + y_0(x_1 - x_0) - x_0(y_1 - y_0)$ =Ax + By + C





$\mathsf{P}=(\mathsf{x},\,\mathsf{y})$

 $V = P - P_0 = (x - x_0, y - y_0)$ $N = \operatorname{Perp}(T) = (y_1 - y_0, -(x_1 - x_0))$

$$P_i = (X_{i, Y_i})$$

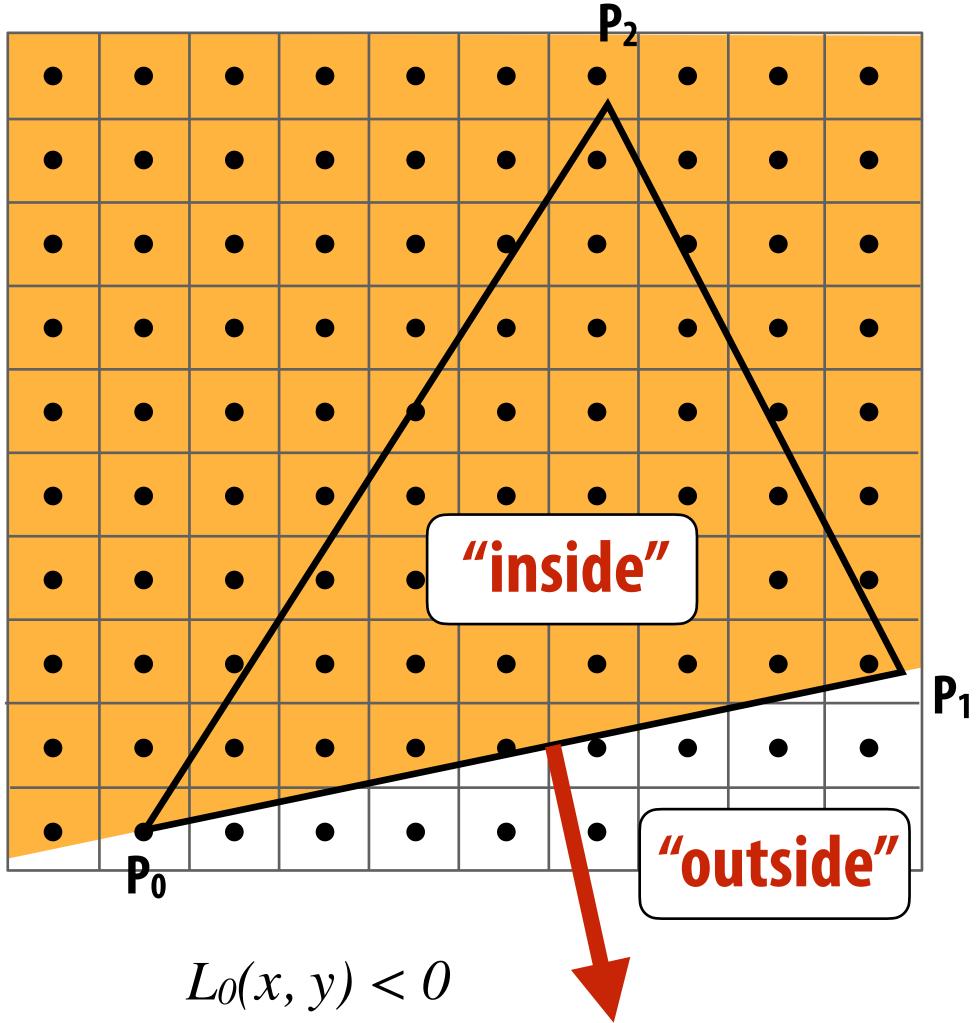
$$A_{i} = dY_{i} = Y_{i+1} - Y_{i}$$

$$B_{i} = dX_{i} = X_{i+1} - X_{i}$$

$$C_{i} = Y_{i}(X_{i+1} - X_{i}) - X_{i}(Y_{i+1} - Y_{i})$$

$$L_i(x, y) = dY_i x - dX_i y + C_i$$

$$L_i(x, y) = 0$$
 : point on edge
> 0 : outside edge
< 0 : inside edge



$$P_i = (X_{i, Y_i})$$

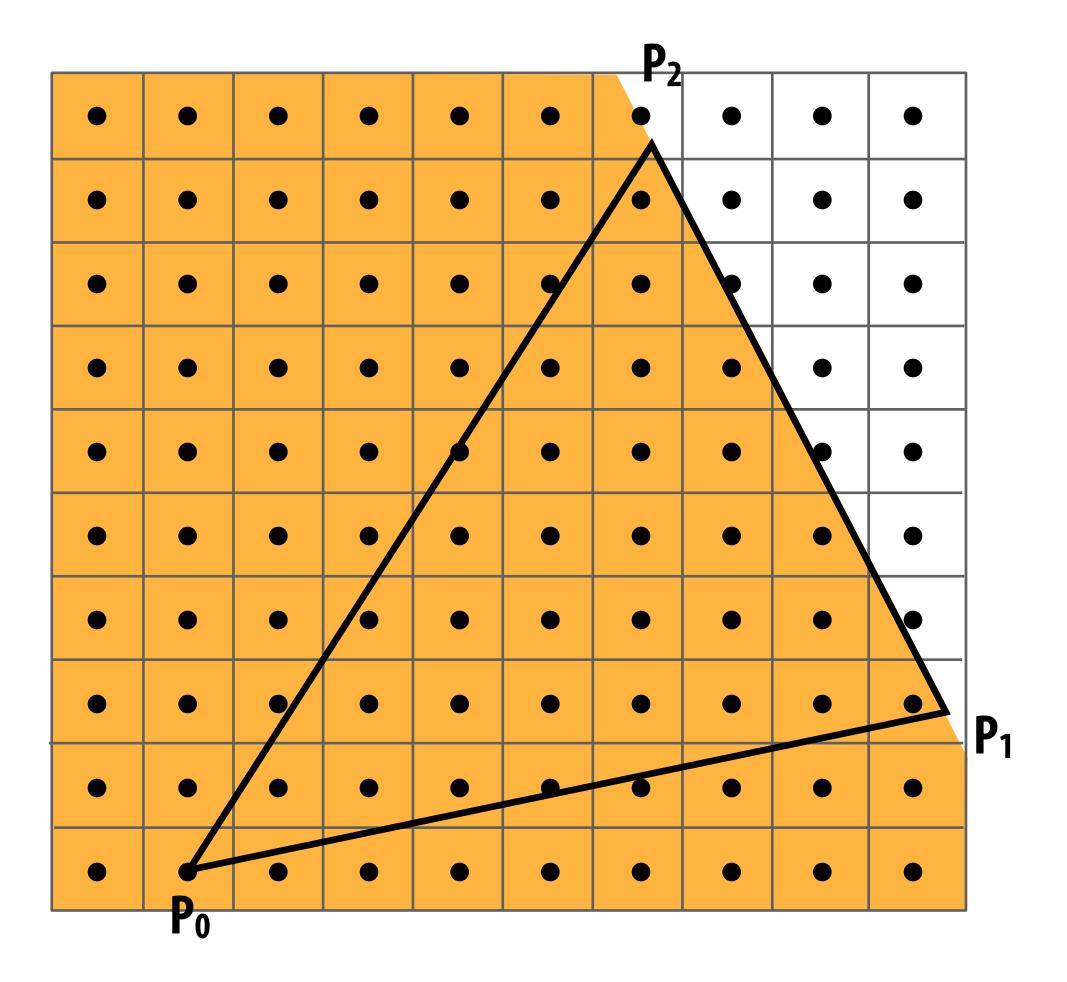
$$A_{i} = dY_{i} = Y_{i+1} - Y_{i}$$

$$B_{i} = dX_{i} = X_{i+1} - X_{i}$$

$$C_{i} = Y_{i}(X_{i+1} - X_{i}) - X_{i}(Y_{i+1} - Y_{i})$$

$$L_i(x, y) = dY_i x - dX_i y + C_i$$

$$L_i(x, y) = 0$$
 : point on edge
> 0 : outside edge
< 0 : inside edge



 $L_l(x, y) < 0$

$$P_i = (X_{i, Y_i})$$

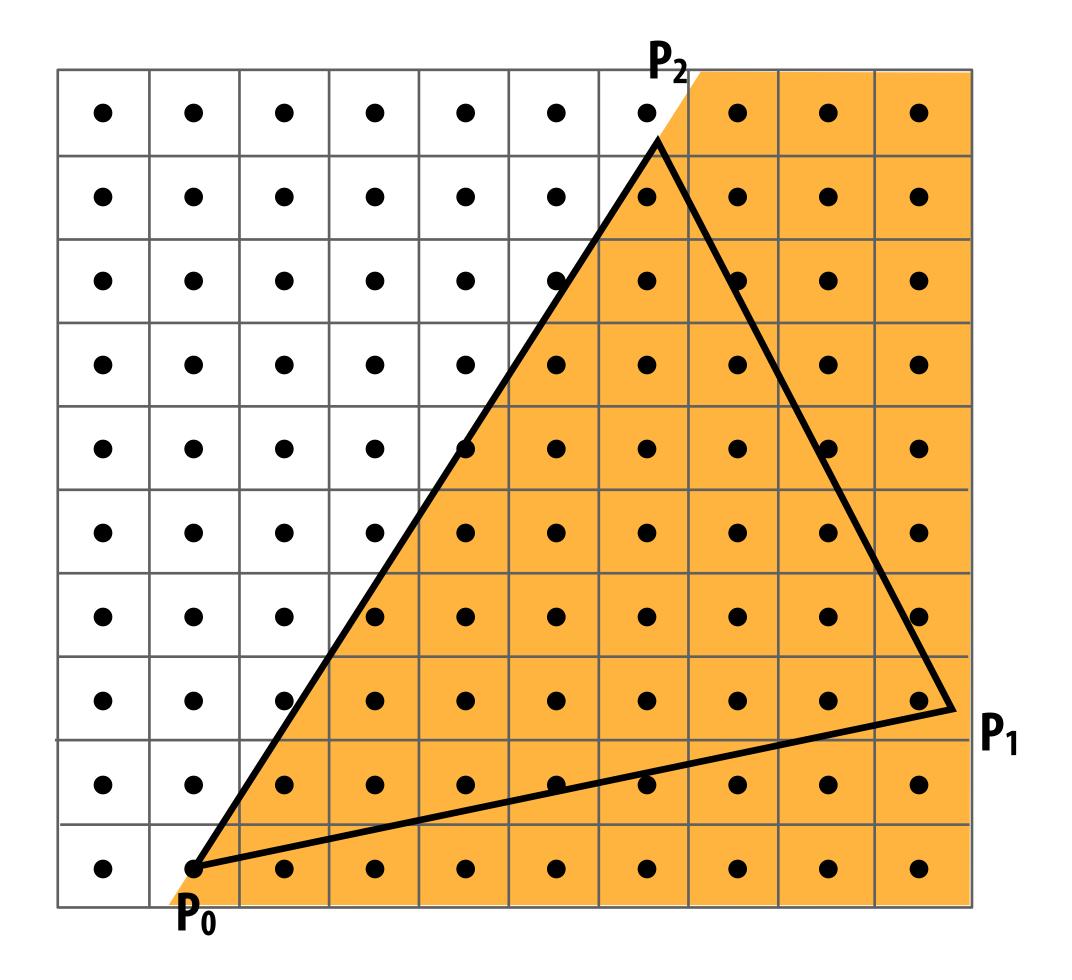
$$A_{i} = dY_{i} = Y_{i+1} - Y_{i}$$

$$B_{i} = dX_{i} = X_{i+1} - X_{i}$$

$$C_{i} = Y_{i}(X_{i+1} - X_{i}) - X_{i}(Y_{i+1} - Y_{i})$$

$$L_i(x, y) = dY_i x - dX_i y + C_i$$

$$L_i(x, y) = 0$$
 : point on edge
> 0 : outside edge
< 0 : inside edge



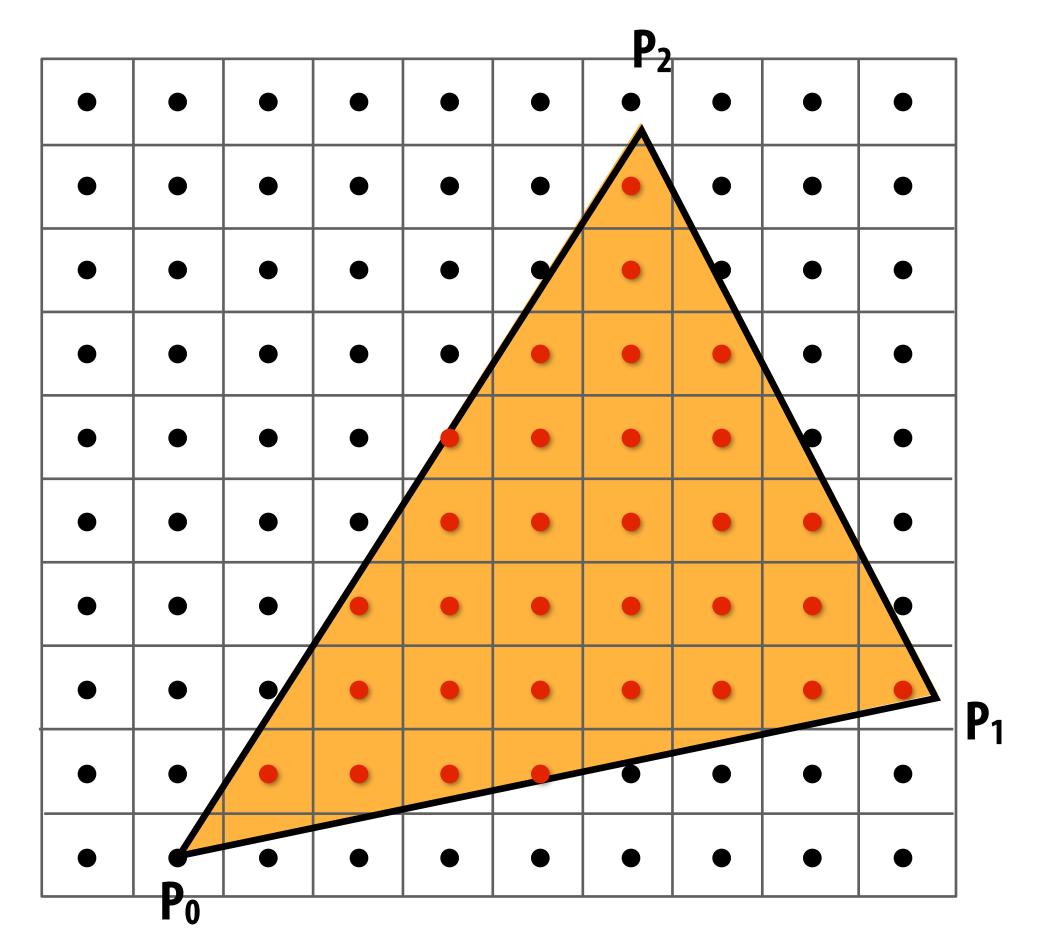
 $L_2(x, y) < 0$

Sample point s = (sx, sy) is inside the triangle if it is inside all three edges.

inside(sx, sy) = $L_0(sx, sy) < 0 \&\&$ $L_1(sx, sy) < 0 \&\&$ $L_2(sx, sy) < 0;$

Note: actual implementation of

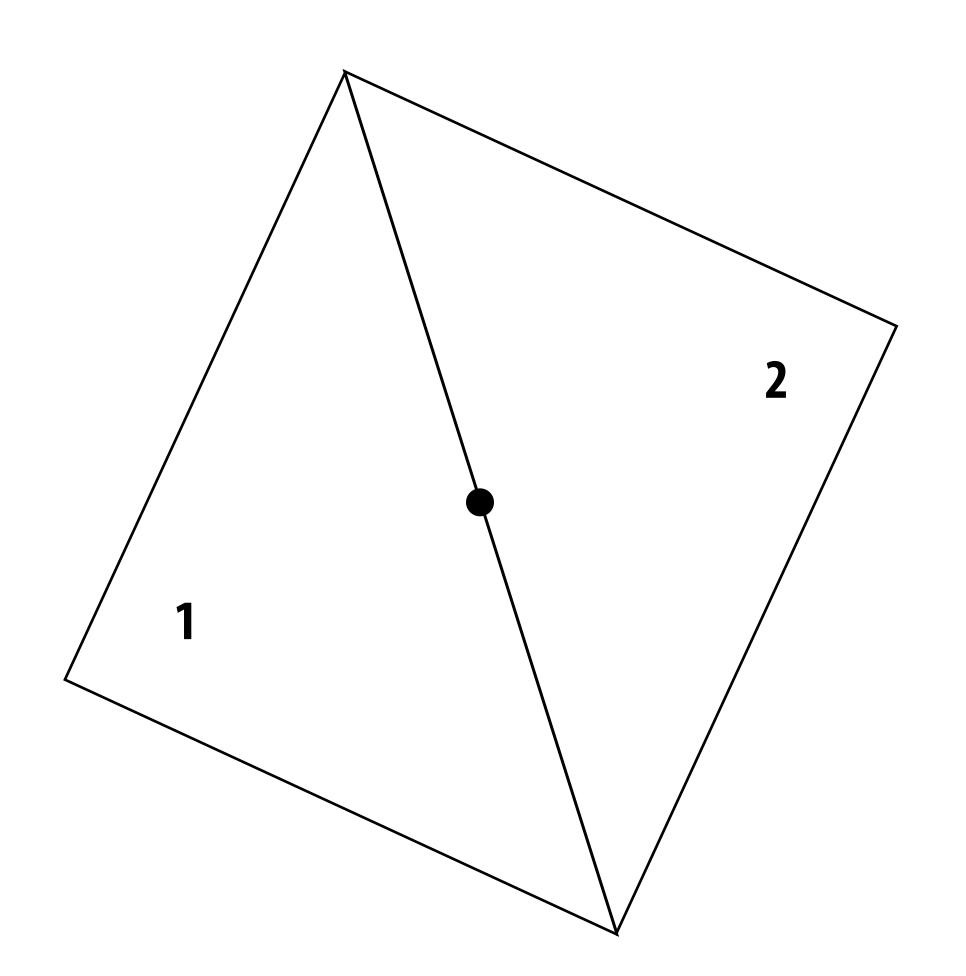
inside(sx,sy) involves \leq checks based on the triangle coverage edge rules (see next slides)



Sample points inside triangle are highlighted red.

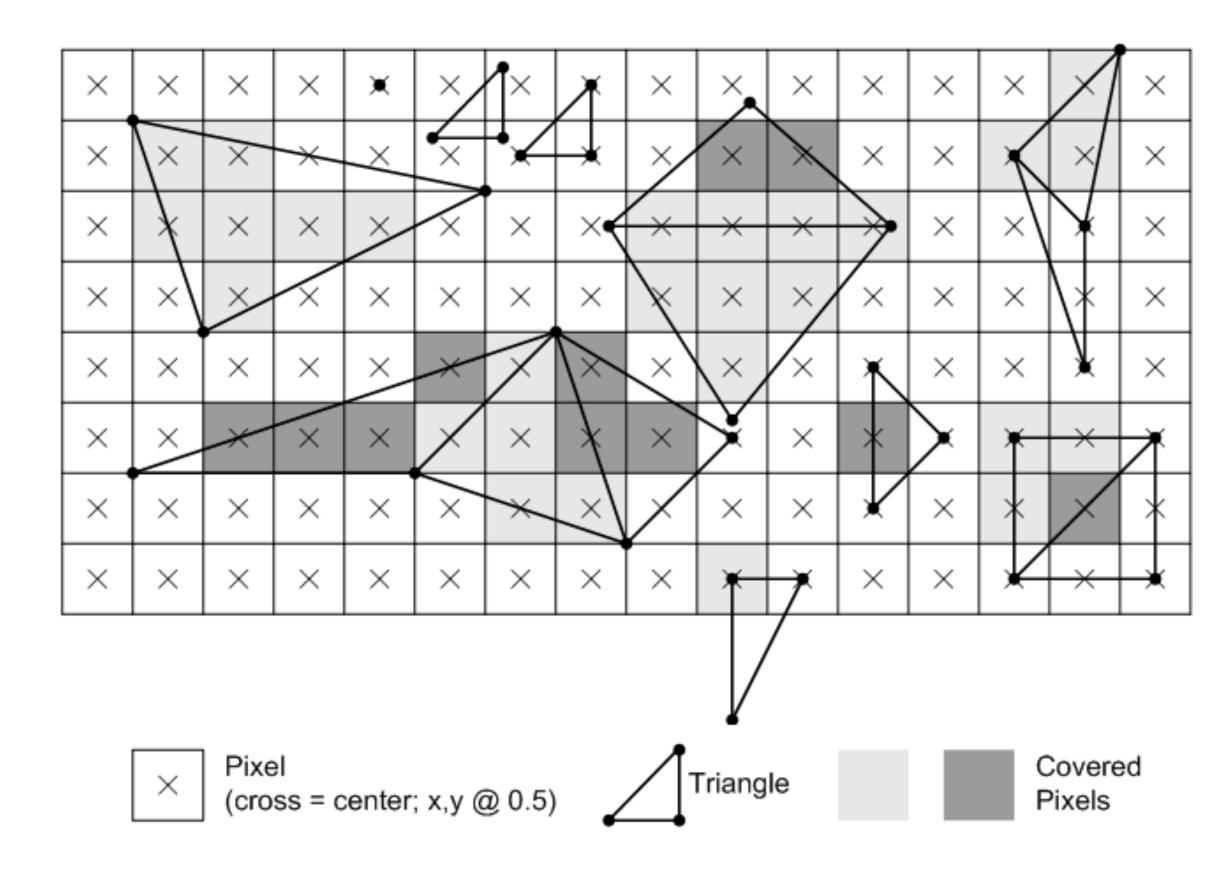
Edge cases (literally)

Is this sample point covered by triangle 1? or triangle 2? or both?



OpenGL/Direct3D edge rules

- When edge falls directly on a screen sample point, the sample is classified as within triangle if the edge is a "top edge" or "left edge"
 - Top edge: horizontal edge that is above all other edges
 - Left edge: an edge that is not exactly horizontal and is on the left side of the triangle. (triangle can have one or two left edges)



Finding covered samples: incremental triangle traversal

 $P_i = (X_i, Y_i)$

$$A_{i} = dY_{i} = Y_{i+1} - Y_{i}$$

$$B_{i} = dX_{i} = X_{i+1} - X_{i}$$

$$C_{i} = Y_{i}(X_{i+1} - X_{i}) - X_{i}(Y_{i+1} - Y_{i})$$

$$L_i(x, y) = dY_i x - dX_i y + C_i$$

$$L_i(x, y) = 0$$
 : point on edge
> 0 : outside edge
< 0 : inside edge

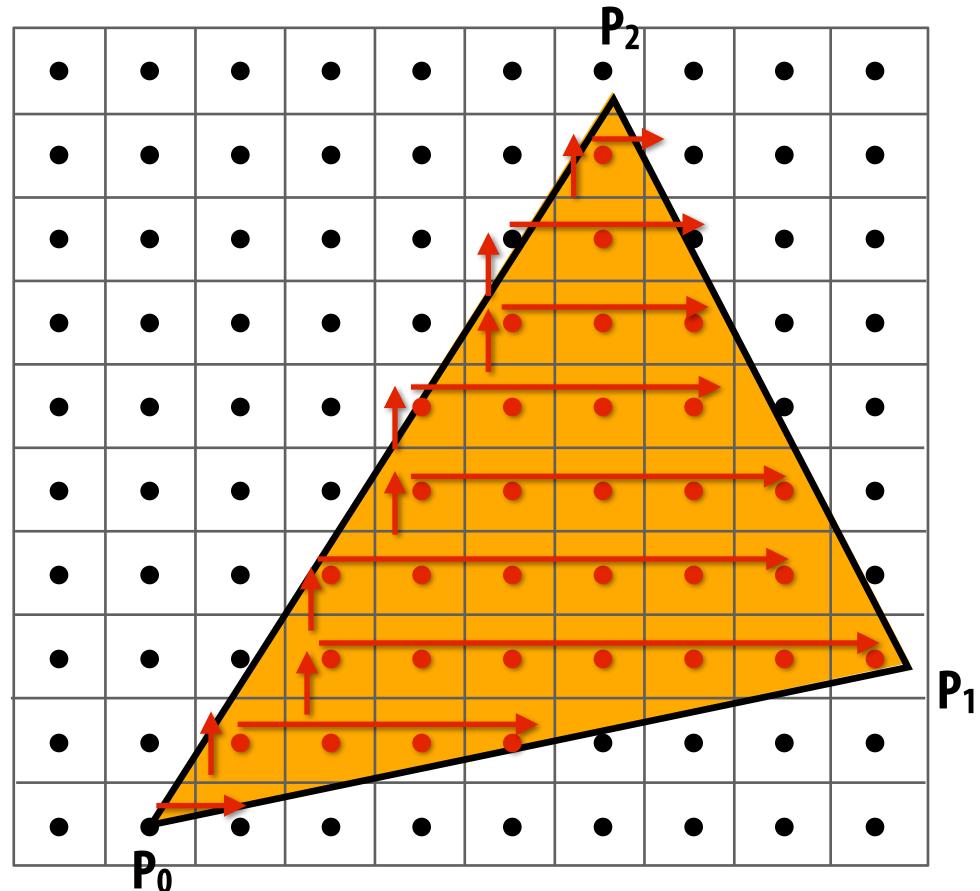
Efficient incremental update:

$$L_{i}(x+1,y) = L_{i}(x,y) + dY_{i} = L_{i}(x,y) + A_{i}$$

$$L_{i}(x,y+1) = L_{i}(x,y) - dX_{i} = L_{i}(x,y) + B_{i}$$

Incremental update saves computation: Only one addition per edge, per sample test

Many traversal orders are possible: backtrack, zig-zag, Hilbert/Morton curves



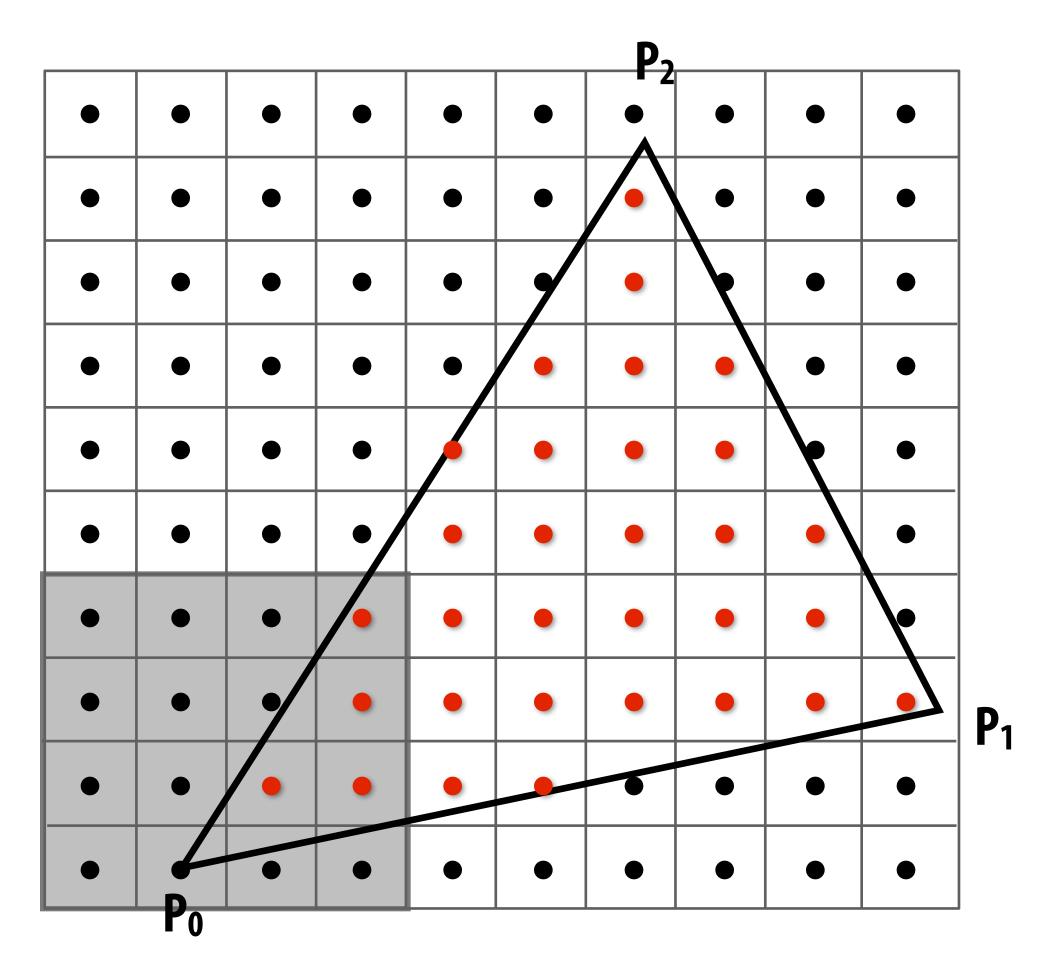
Modern approach: tiled triangle traversal

Traverse triangle in blocks

Test all samples in block against triangle in parallel

Advantages:

- Simplicity of parallel execution overcomes cost of extra point-in-triangle tests (most triangles are big enough to cover many samples)
- Can skip sample testing work: entire block not in triangle ("early out"), entire block entirely within triangle ("early in")
- Additional advantages related to accelerating occlusion computations (not discussed today)



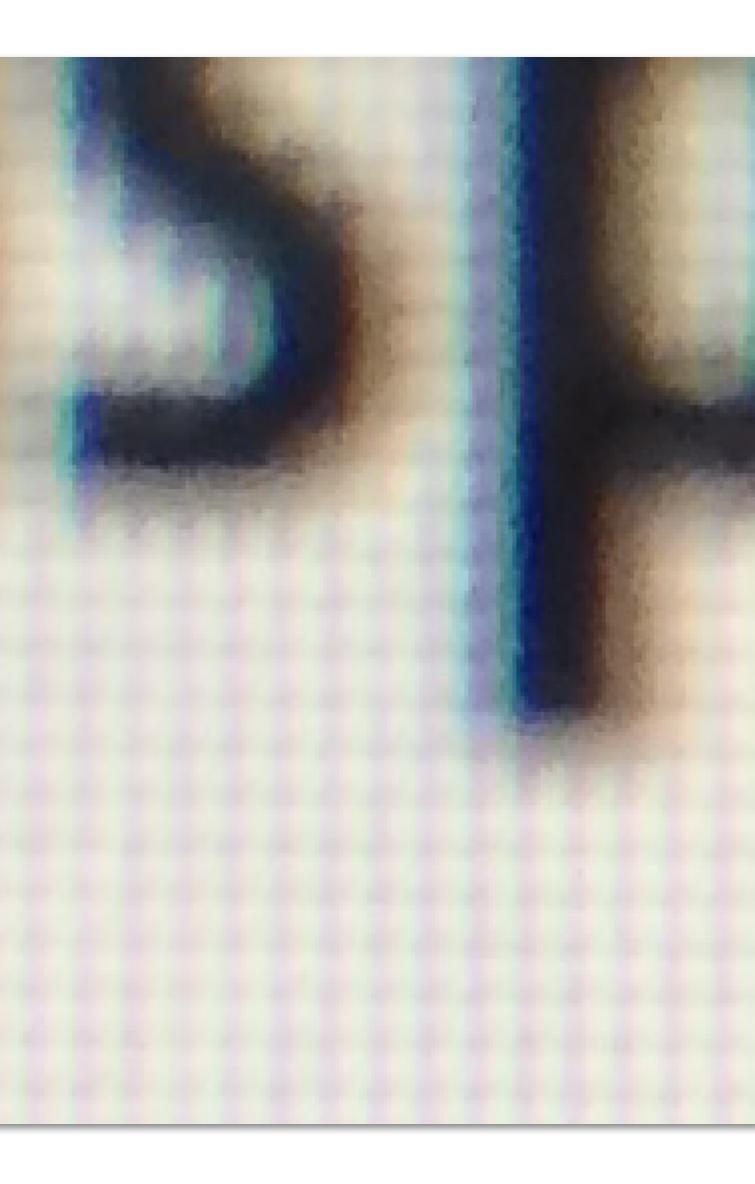
All modern graphics processors (GPUs) have special-purpose hardware for efficiently performing point-in-triangle tests

Recall: pixels on a screen

Each image sample sent to the display is converted into a little square of light of the appropriate color: (a pixel = picture element)

LCD display pixel on my laptop

* Thinking of each LCD pixel as emitting a square of uniform intensity light of a single color is a bit of an approximation to how real displays work, but it will do for now.

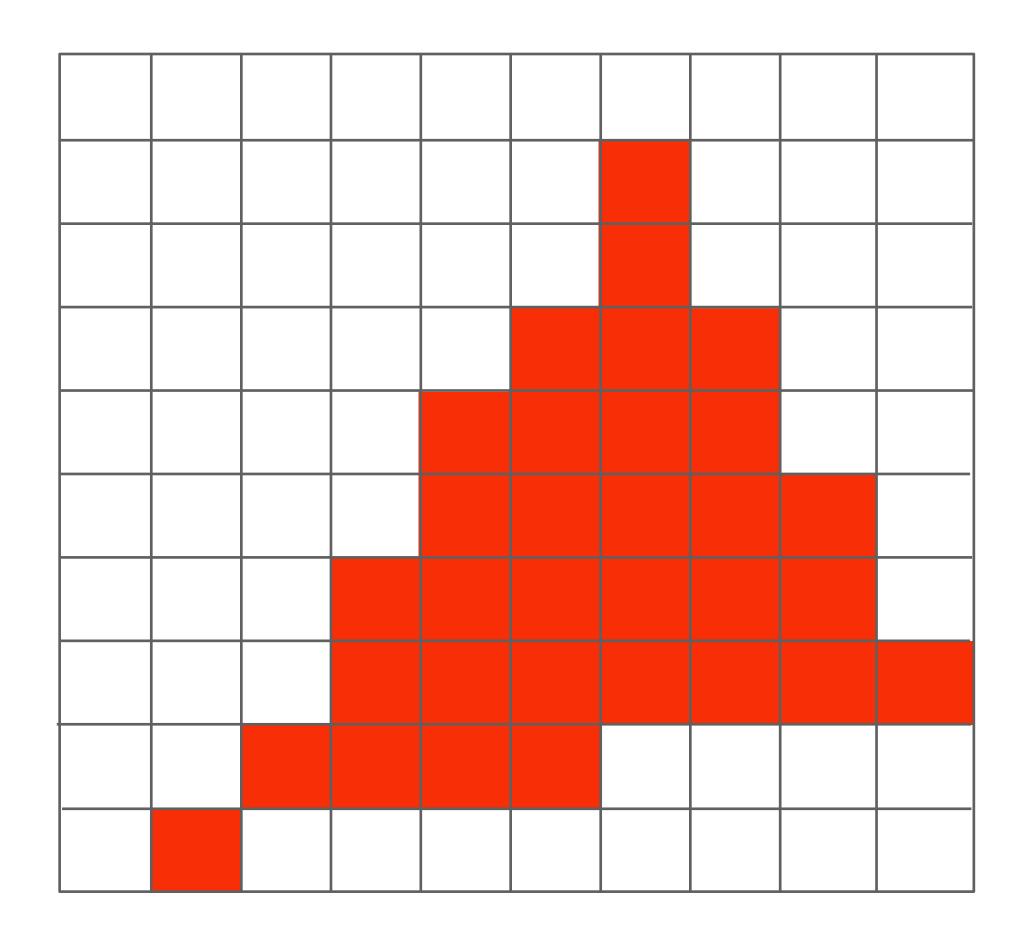


So, if we send the display this sampled signal

0	0	0	0	0	0	0	0
0	0	0	0	0	0		0
0	0	0	0	0	0		0
0	0	0	0	0			
0	0	0	0				
0	0	0	0				
0	0	0					
0	0	0					
0	0					0	0
0		0	0	0	0	0	0

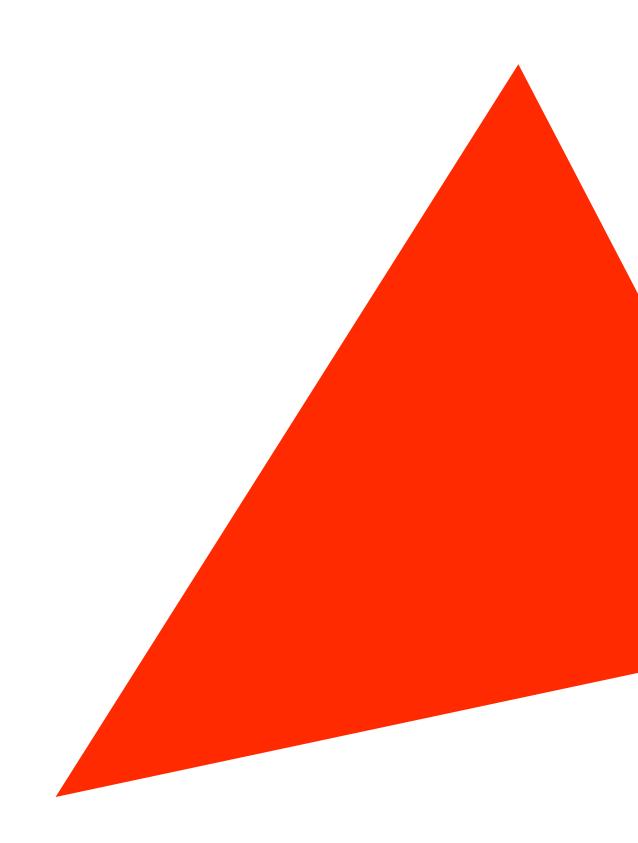
- 0 0
- 0 0
- 0 0
- 0 0
- 0 0
- 0
- 0
- • • •
- 0 0

The display physically emits this signal



Given our simplified "square pixel" display assumption, we've effectively performed a piecewise constant reconstruction

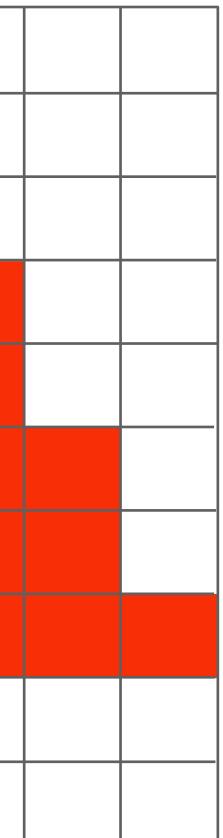
Compare: the continuous triangle function



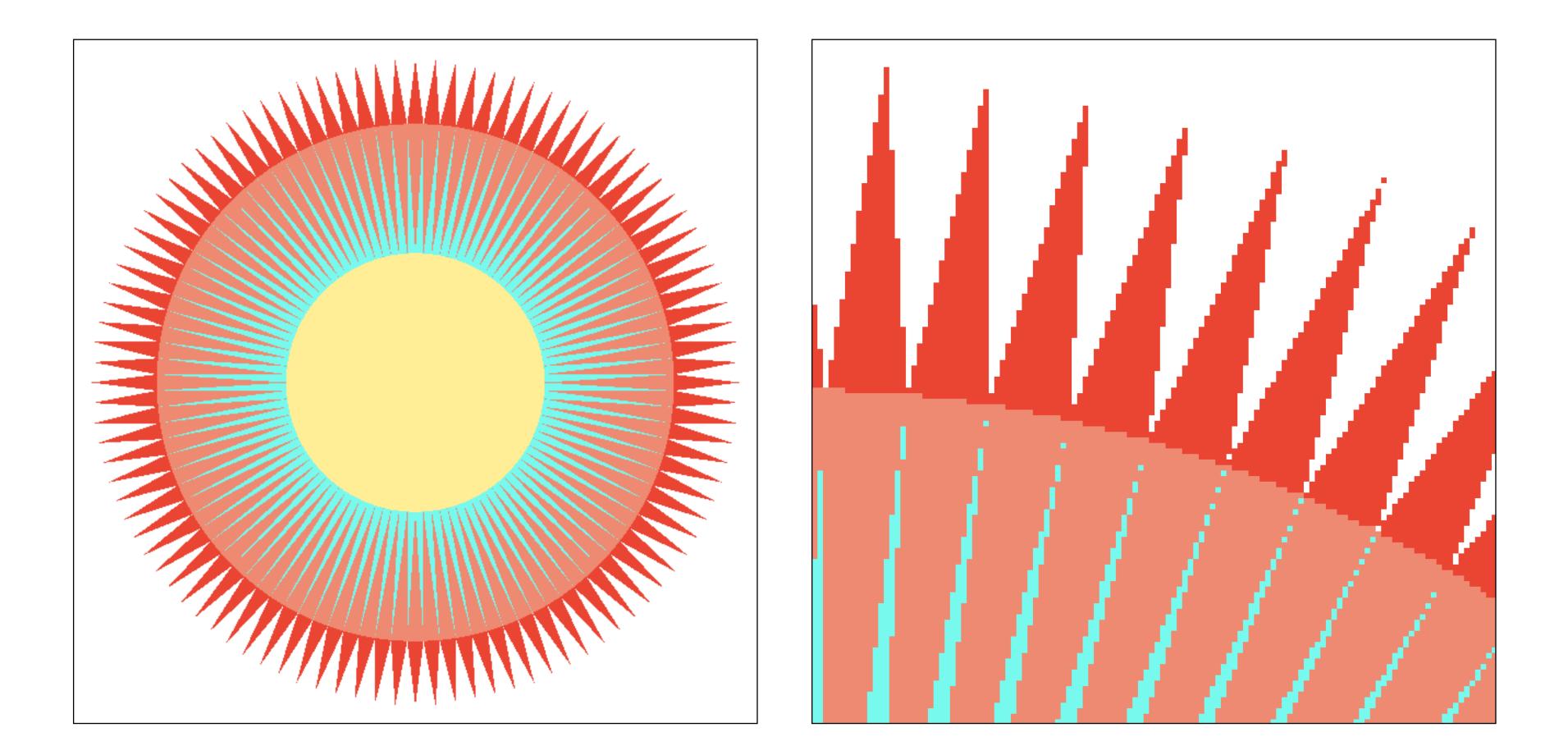
What's wrong with this picture?





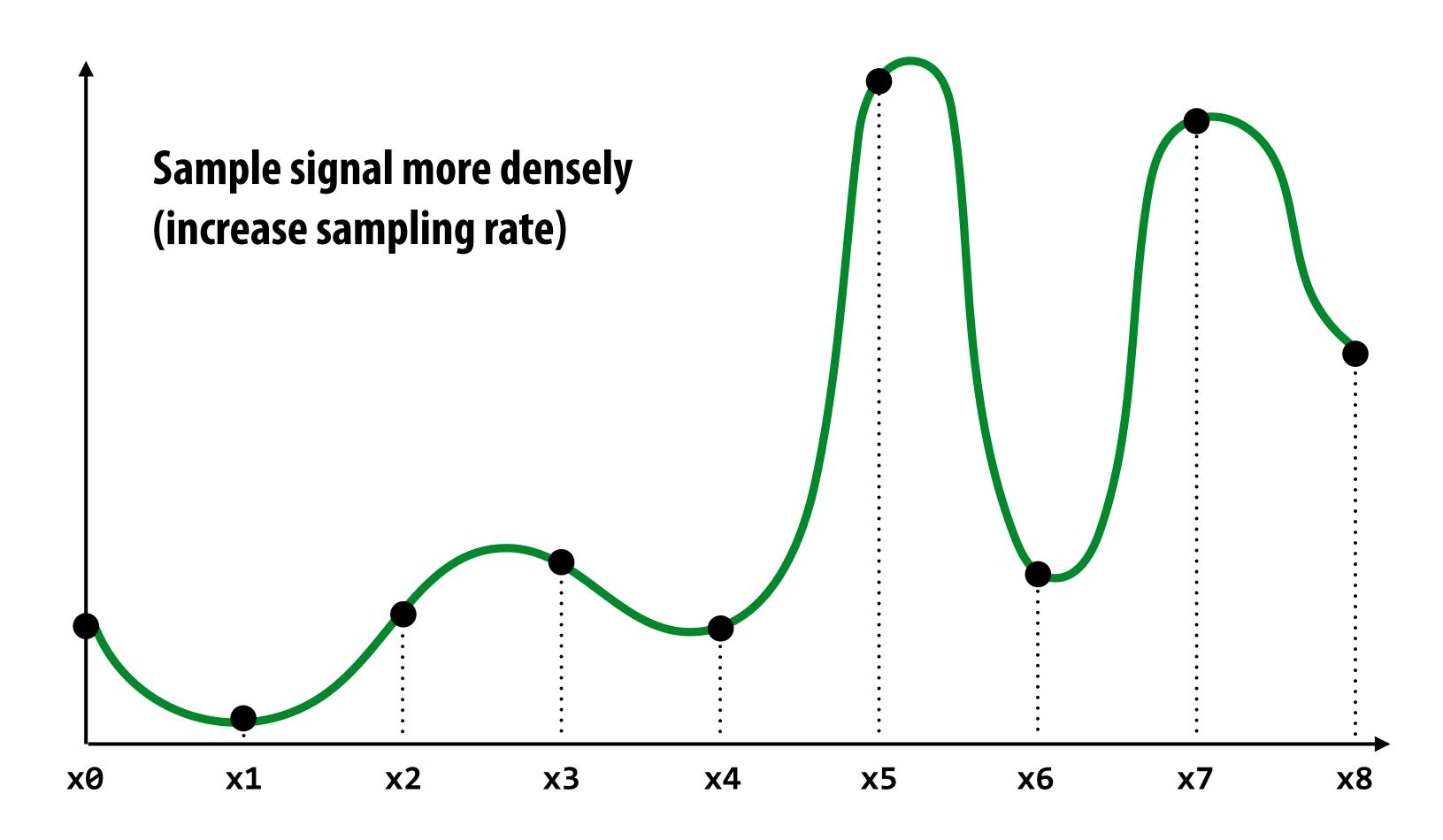


Jaggies (staircase pattern)

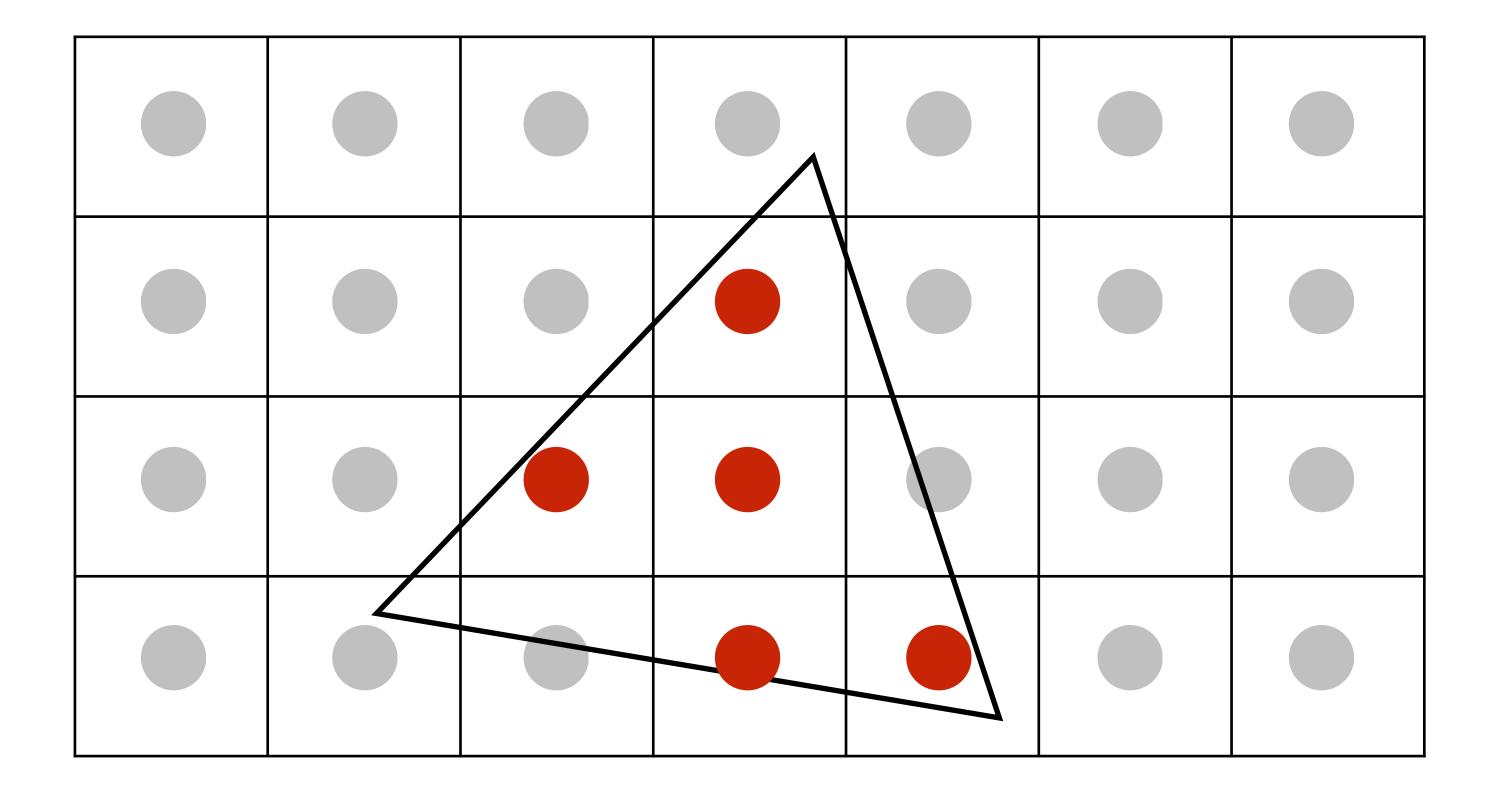


Is this the best we can do?

Reminder: how can we represent a sampled signal more accurately?

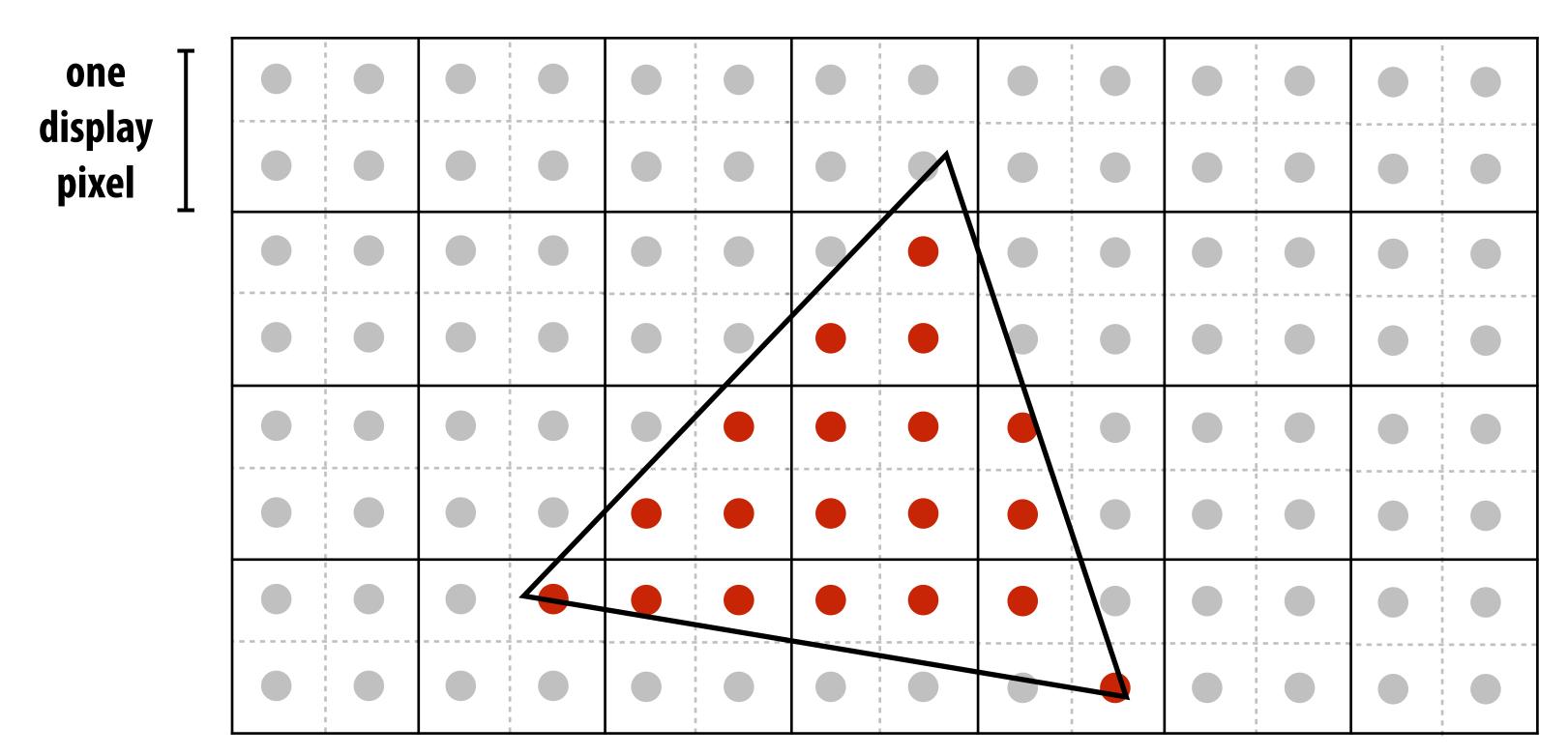


Point sampling: one sample per pixel



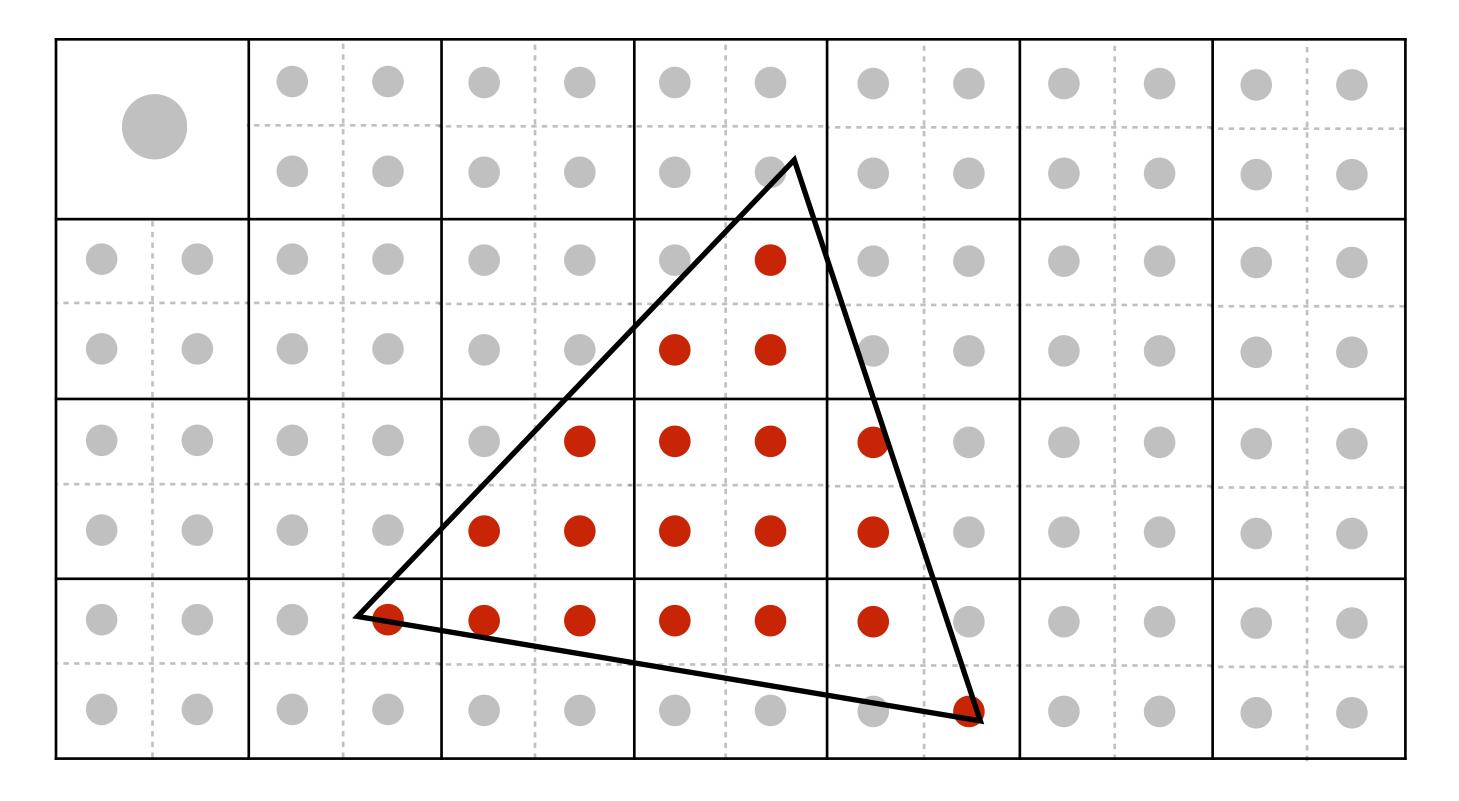
Take N x N samples in each pixel

(but... how do we use these samples to drive a display, since there are four times more samples than display pixels!)



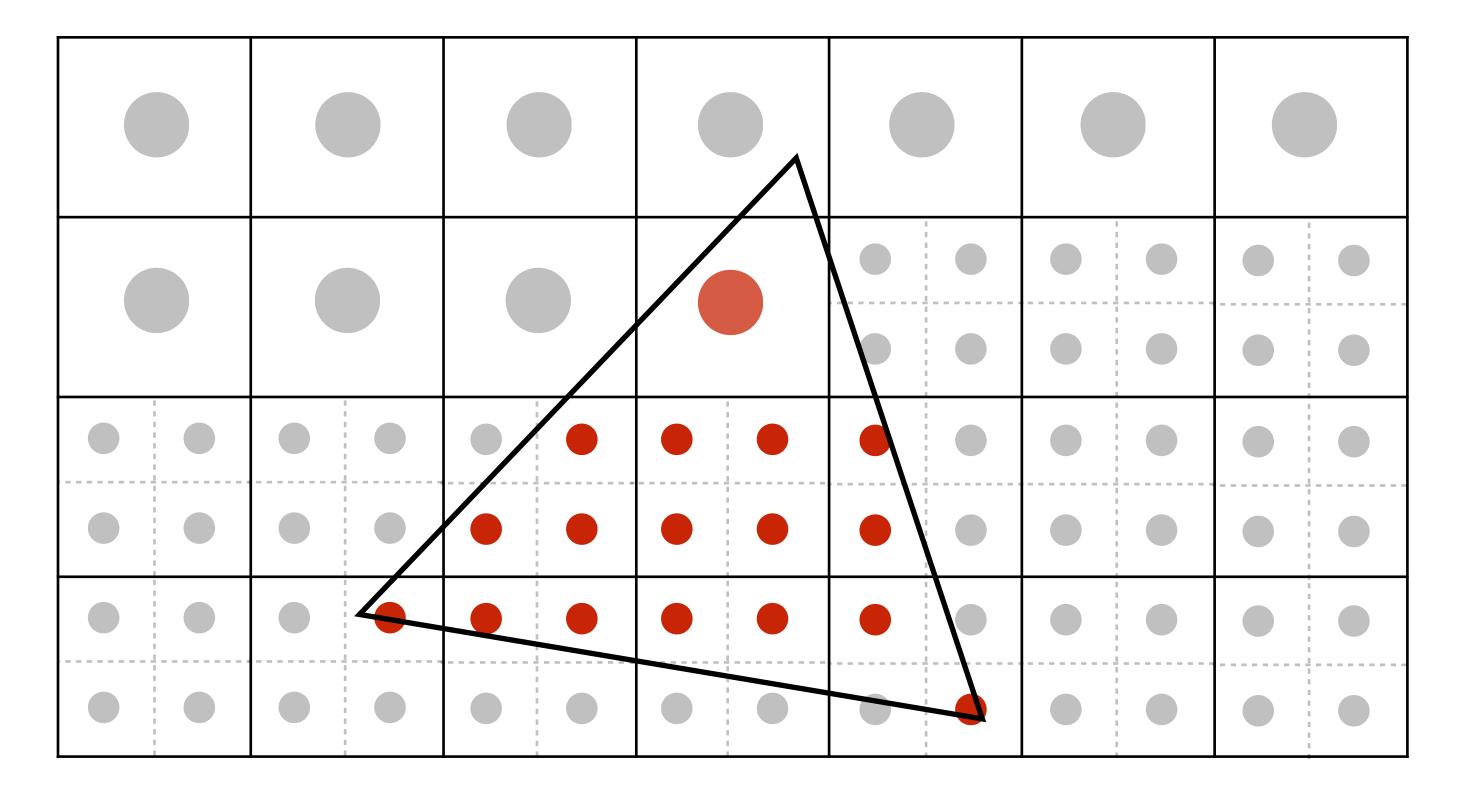
2x2 supersampling

Average the N x N samples "inside" each pixel



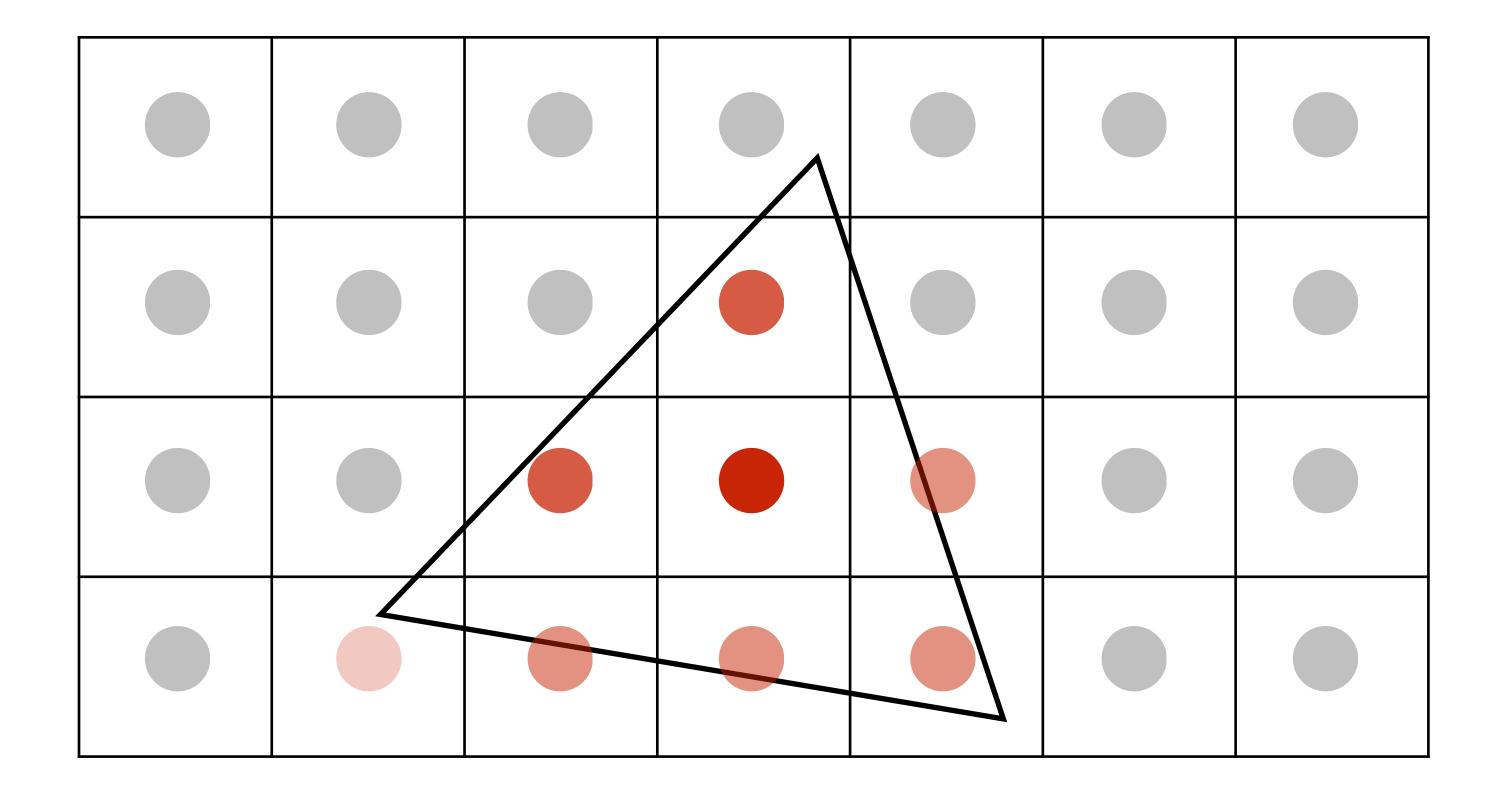
Averaging down

Average the N x N samples "inside" each pixel



Averaging down

Average the N x N samples "inside" each pixel

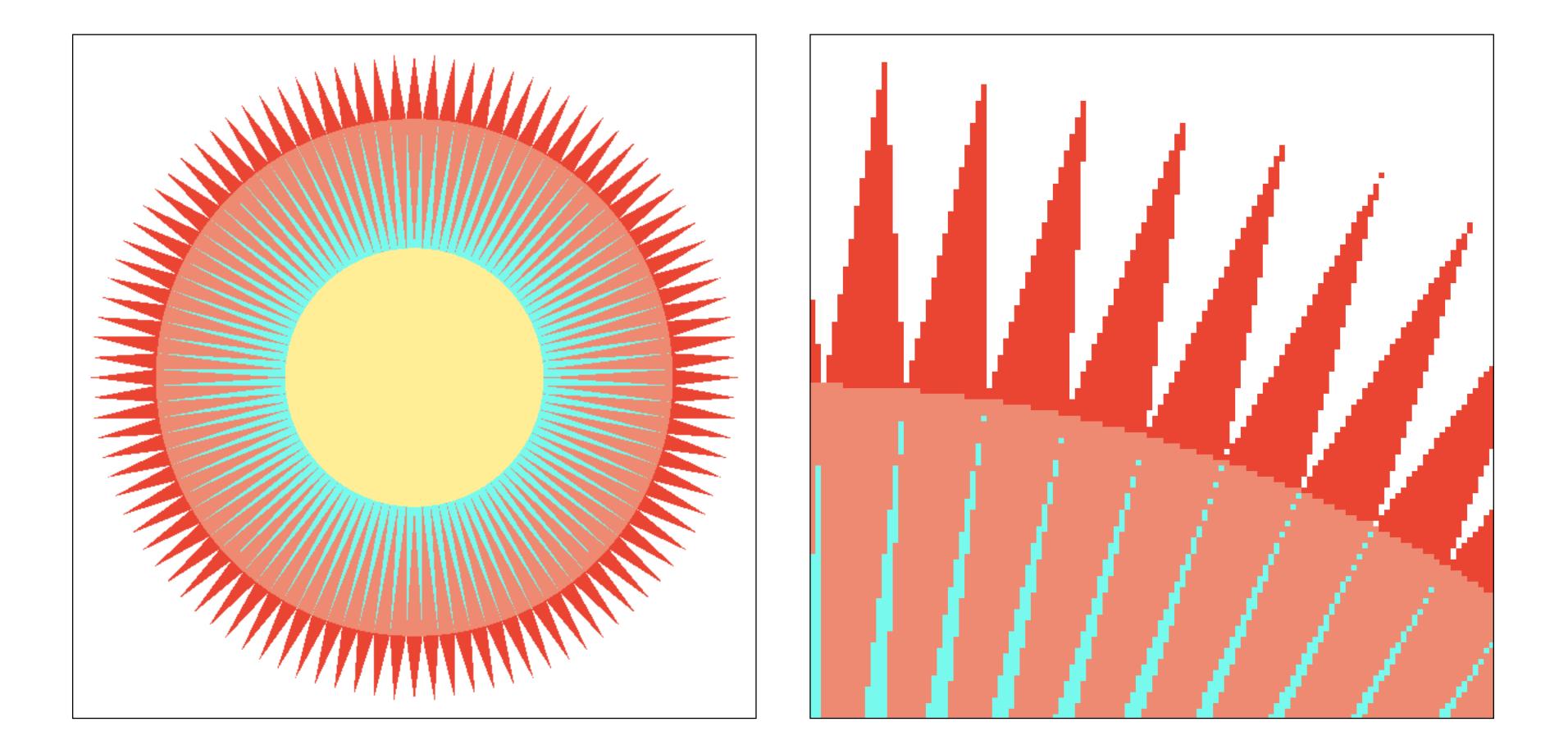


Supersampling: result

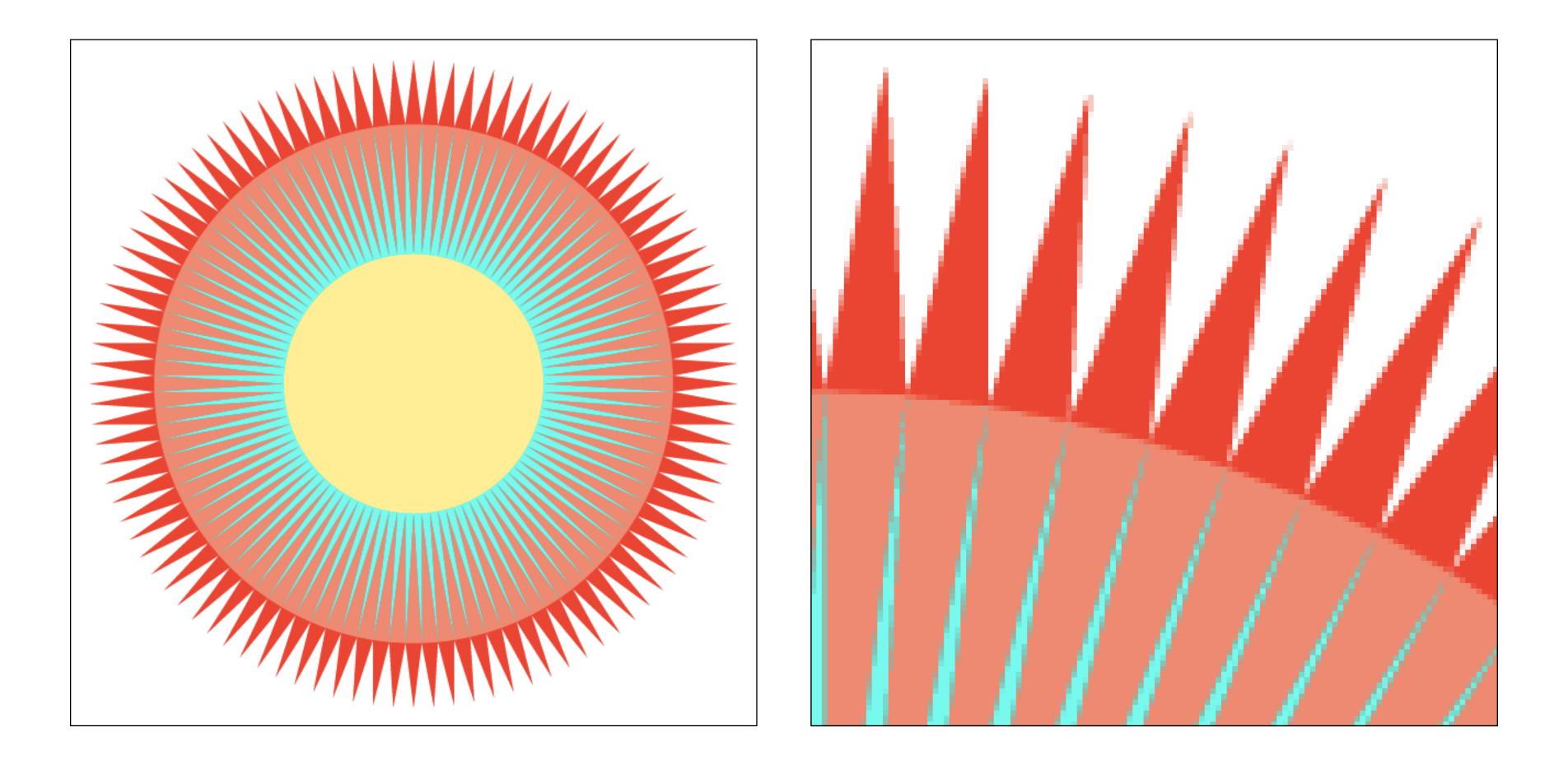
This is the corresponding signal emitted by the display

		75%		
	100%	100%	50%	
25%	50%	50%	50%	

Images rendered using one sample per pixel



4x4 supersampling + downsampling

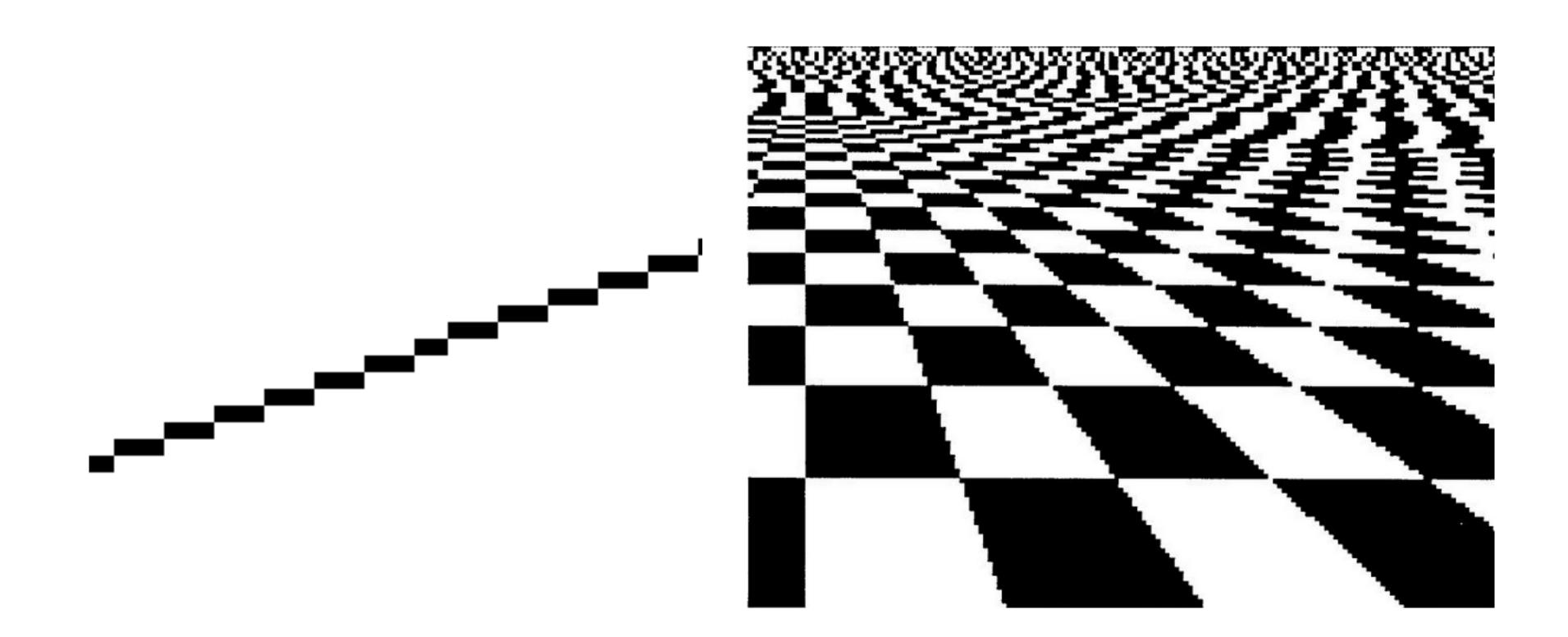


Pixel value is average of 4x4 samples per pixel

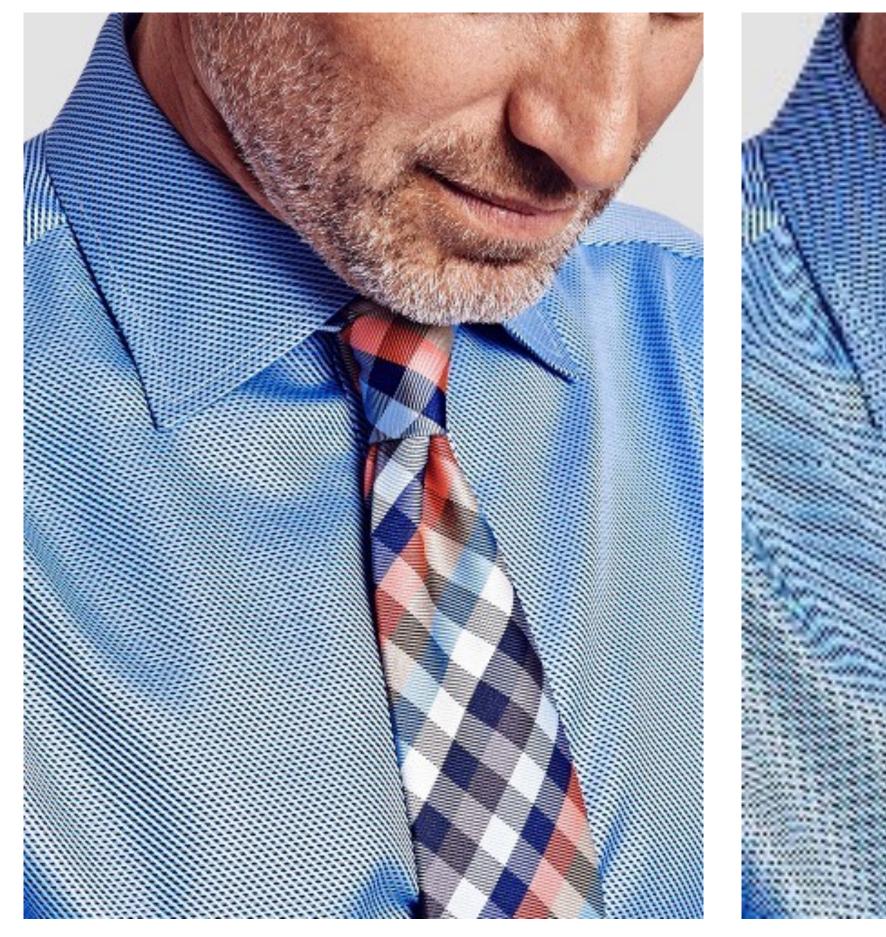
Let's understand what just happened in a more principled way

More examples of sampling artifacts in computer graphics

Jaggies (staircase pattern)



Moiré patterns in imaging

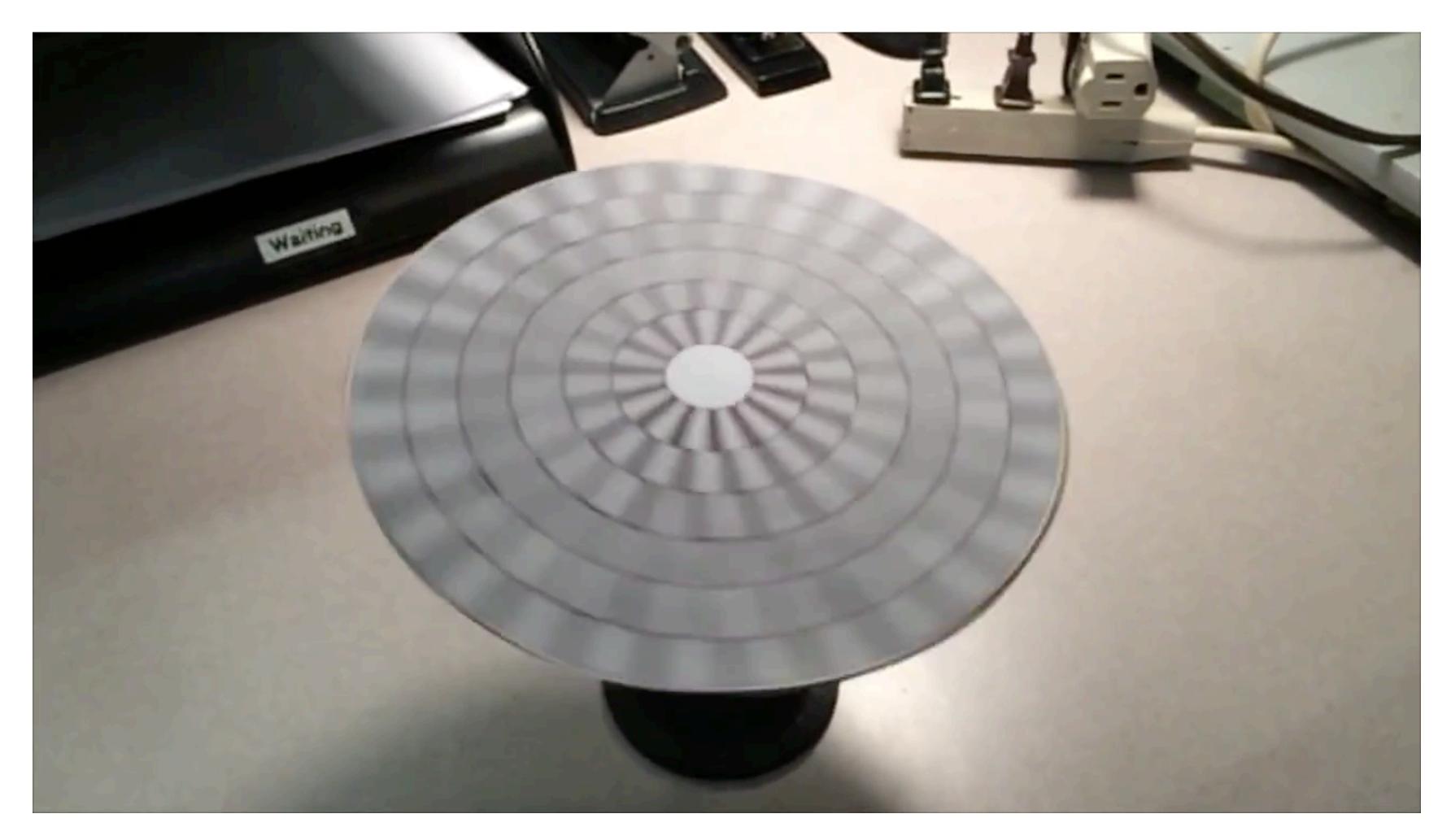


Full resolution image



1/2 resolution image: skip pixel odd rows and columns

Wagon wheel illusion (false motion)



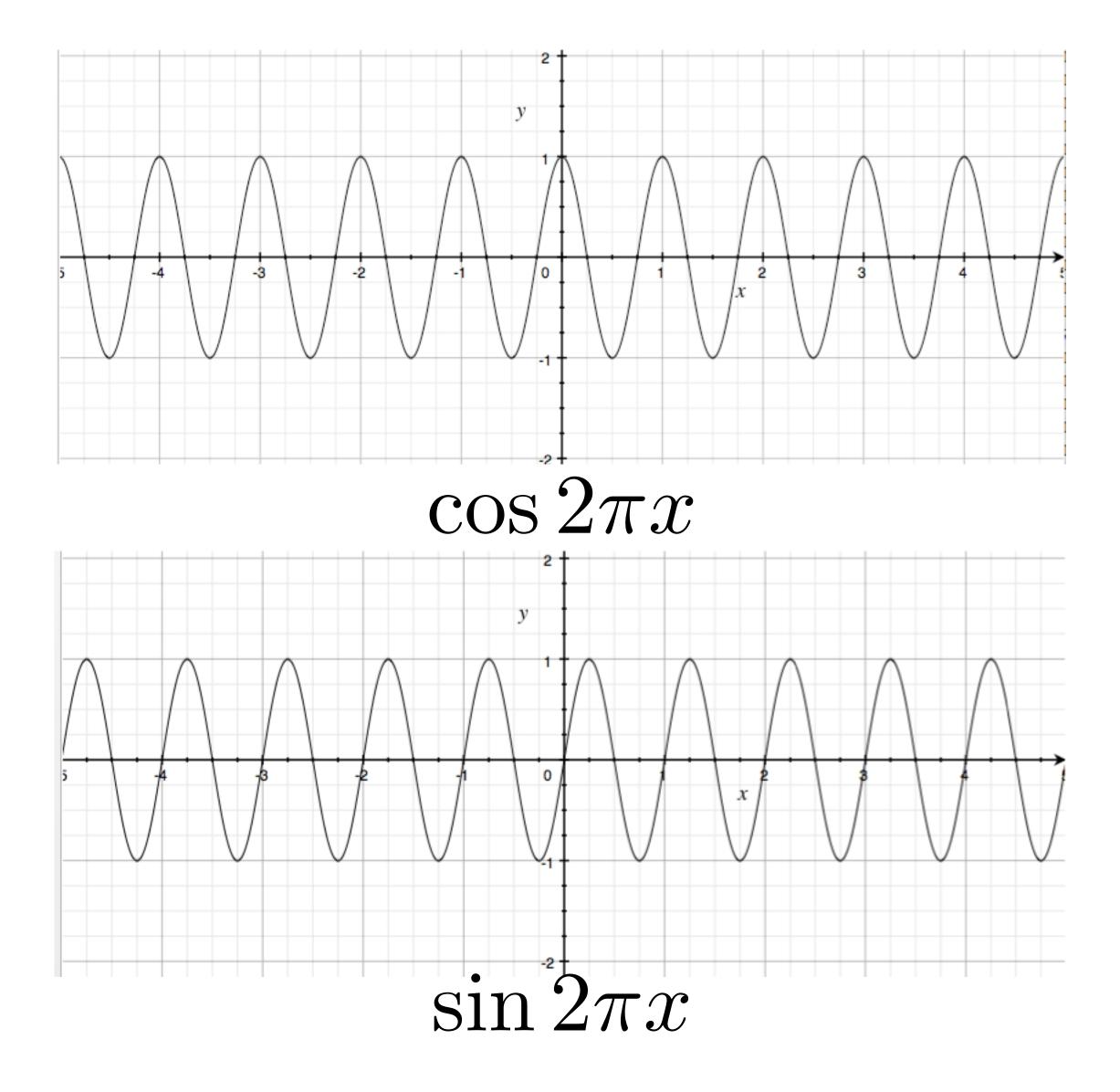
Camera's frame rate (temporal sampling rate) is too low for rapidly spinning wheel.

Created by Jesse Mason, https://www.youtube.com/watch?v=QOwzkND_ooU

Sampling artifacts in computer graphics

- **Artifacts due to sampling "Aliasing"**
 - Jaggies sampling in space
 - Wagon wheel effect sampling in time
 - Moire undersampling images (and texture maps)
 - [Many more] ...
- We notice this in fast-changing signals, when we sample the signal too sparsely

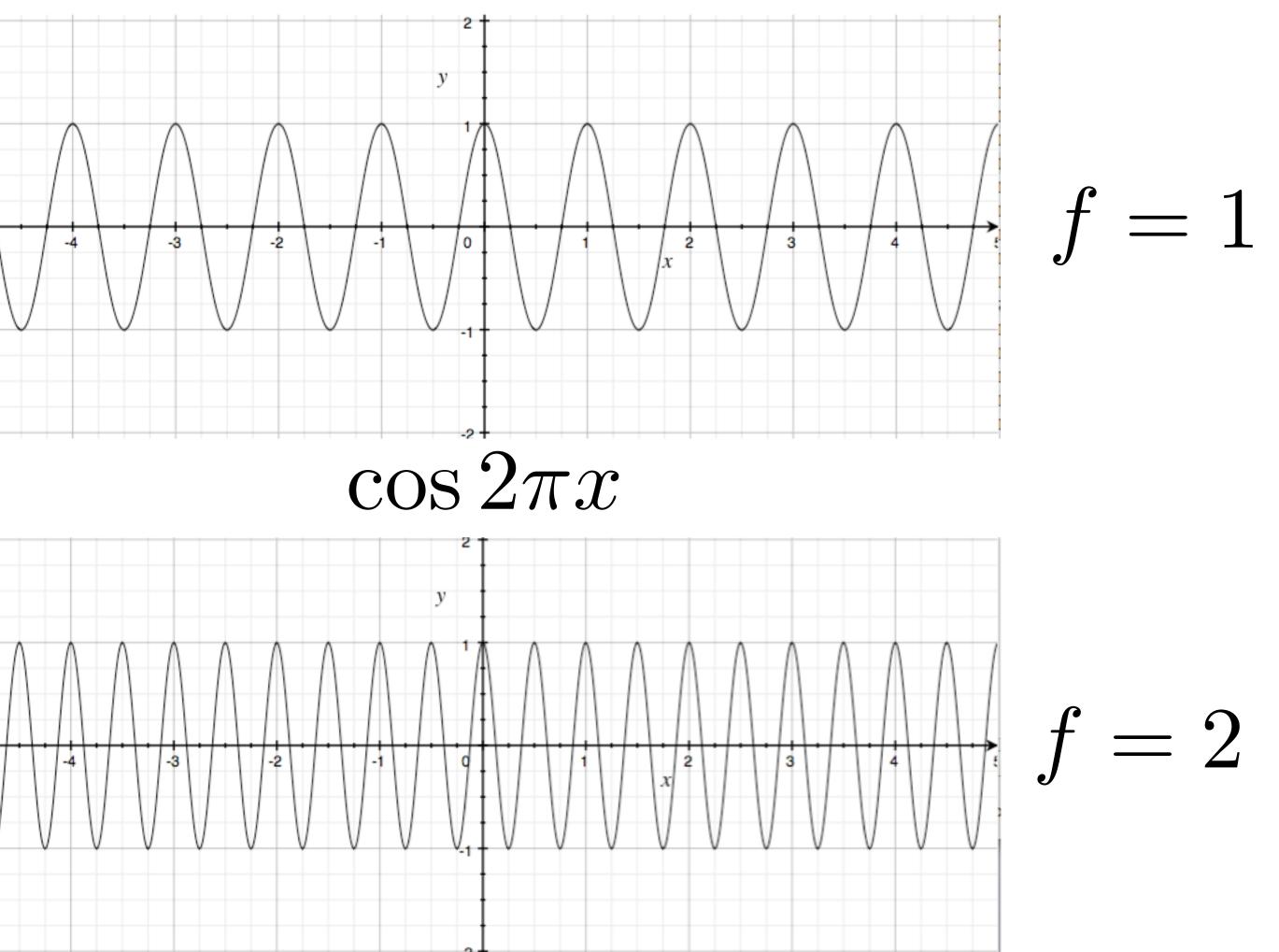
Sines and cosines

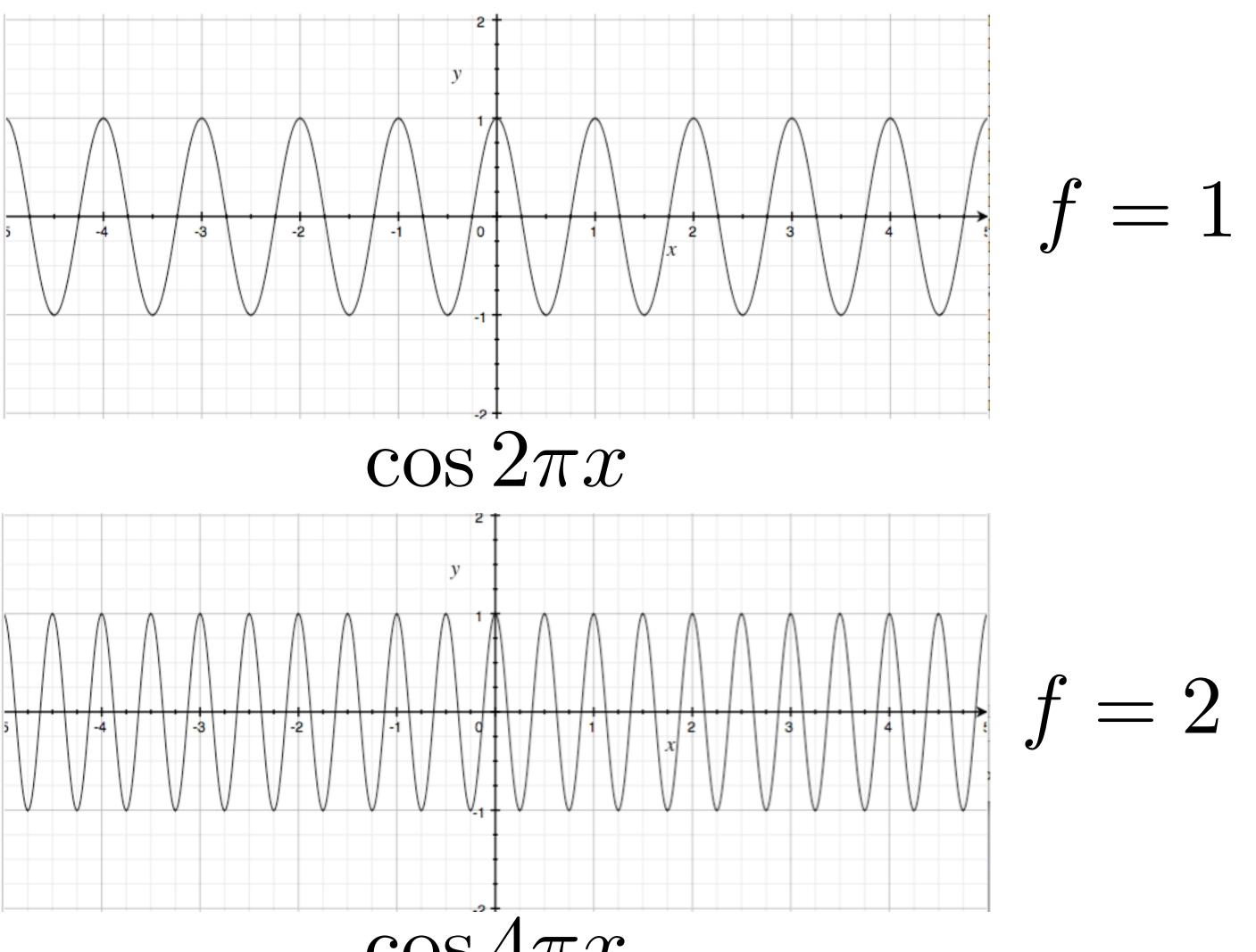


Frequencies

 $\cos 2\pi f x$

 $f = \frac{1}{T}$





 $\cos 4\pi x$

Representing sound wave as a superposition of frequencies

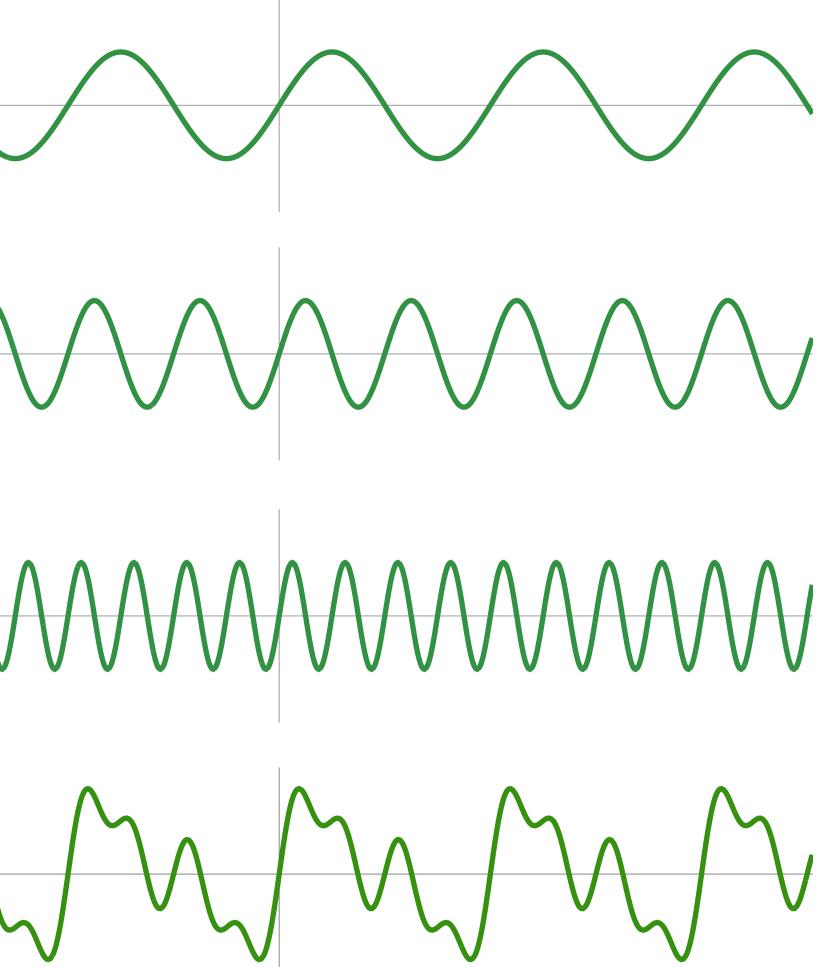
$$f_1(x) = sin(\pi x)$$

$$f_2(x) = sin(2\pi x)$$

$$f_4(x) = sin(4\pi x)$$

 $f(x) = 1.0 f_1(x) + 0.75 f_2(x) + 0.5 f_4(x)$





Audio spectrum analyzer: representing sound as a sum of its constituent frequencies

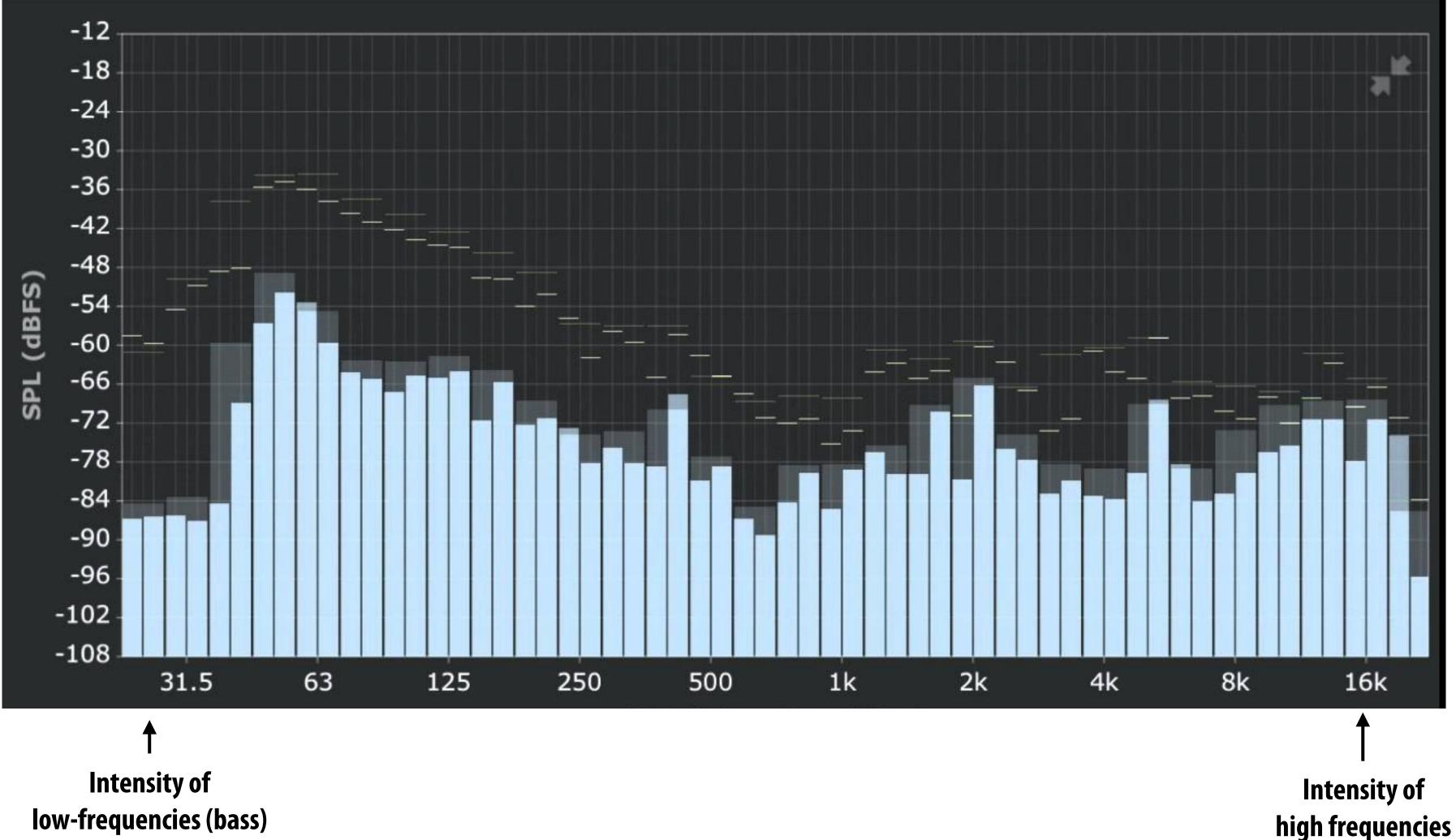


Image credit: ONYX Apps

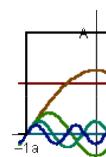
How to compute frequency-domain representation of a signal?

Fourier transform

Represent a function as a weighted sum of sines and cosines

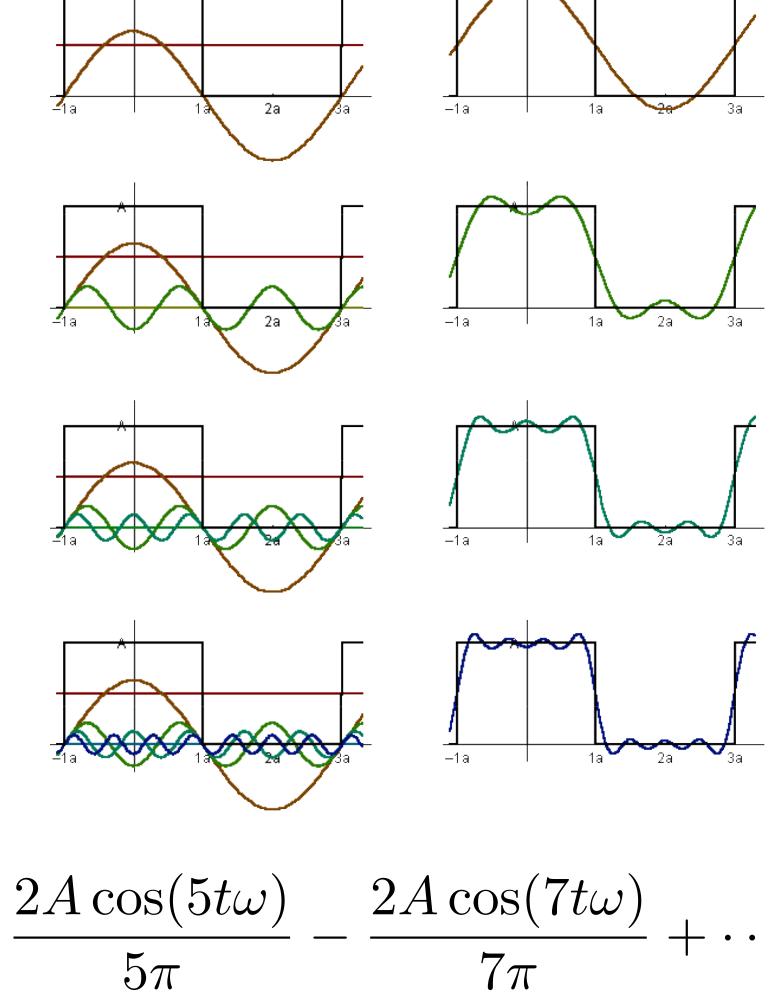






Joseph Fourier 1768 - 1830

 $f(x) = \frac{A}{2}$ $2A\cos(t\omega) = 2A\cos(3t\omega)$ 3π π



Fourier transform

Convert representation of signal from primal domain (spatial/ temporal) to frequency domain by projecting signal into its **component frequencies**

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

• 2D form:

$$F(u,v) = \int \int f(x,y) e^{-2\pi i (x,y)} e^{-2\pi i (x,$$

Recall:
$$e^{ix} = \cos x + i \sin x$$

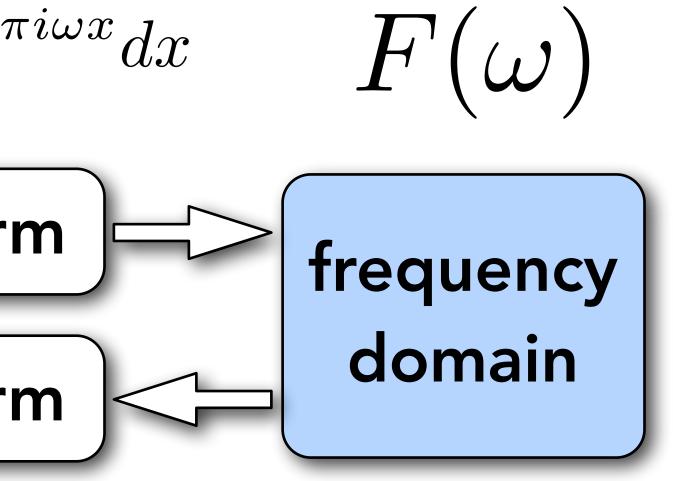
 $-i\sin(2\pi\omega x))dx$

(ux+vy)dxdy

Fourier transform decomposes a signal into its constituent frequencies

$$f(x) F(\omega) = \int_{-\infty}^{\infty} f(x)e^{-2\pi x}$$
Spatial Fourier transformed on the second second

$$f(x) = \int_{-\infty}^{\infty} F(\omega) e^{2\pi}$$



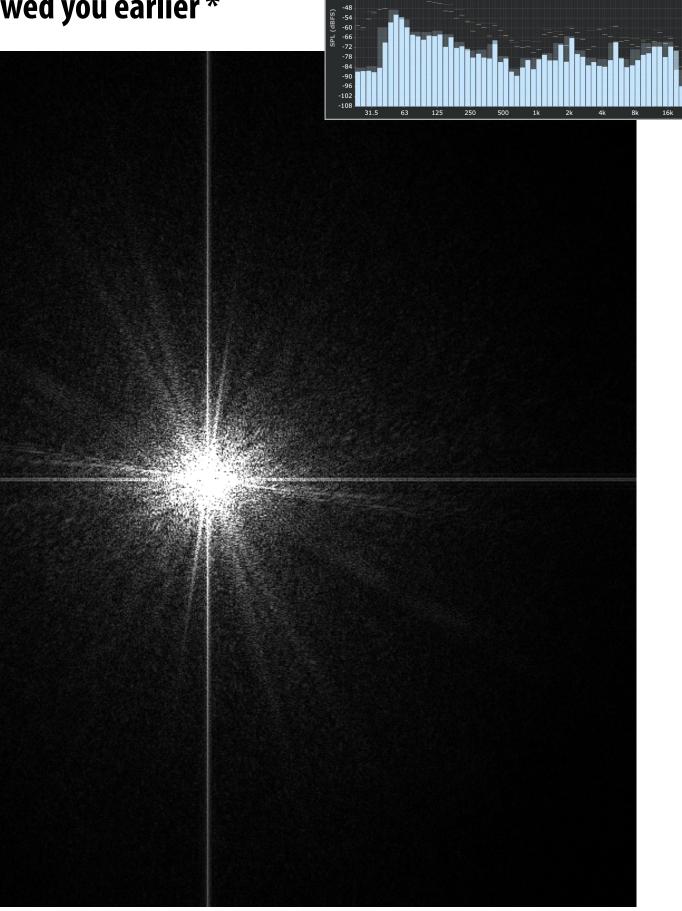
 $^{\cdot i\omega x}d\omega$

Visualizing the frequency content of images

Visualization below is the 2D frequency domain equivalent of the 1D audio spectrum I showed you earlier *

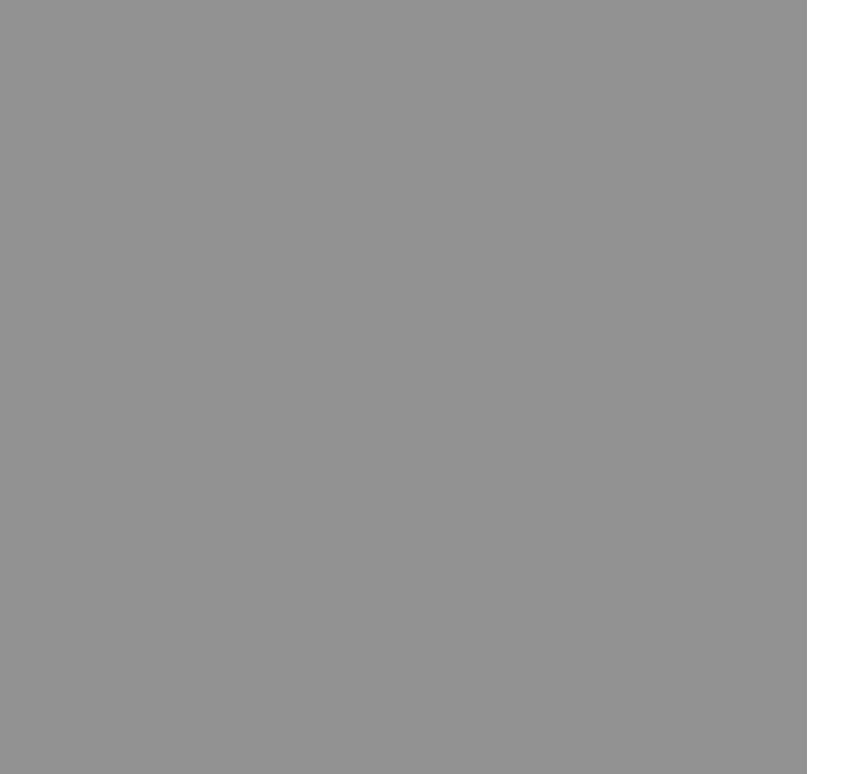


Spatial domain result

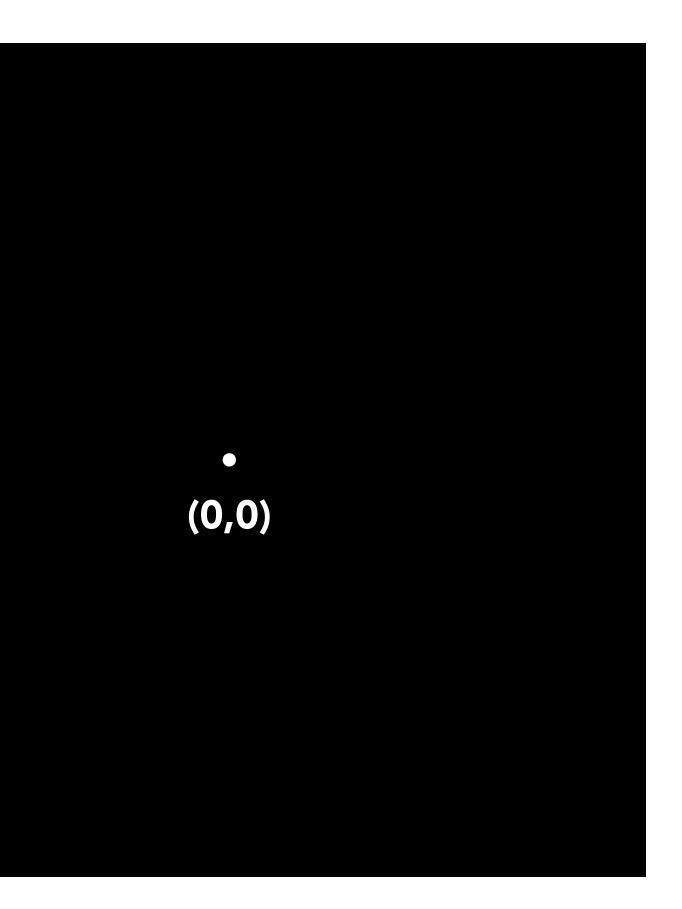


Spectrum

Constant signal (in primal domain)

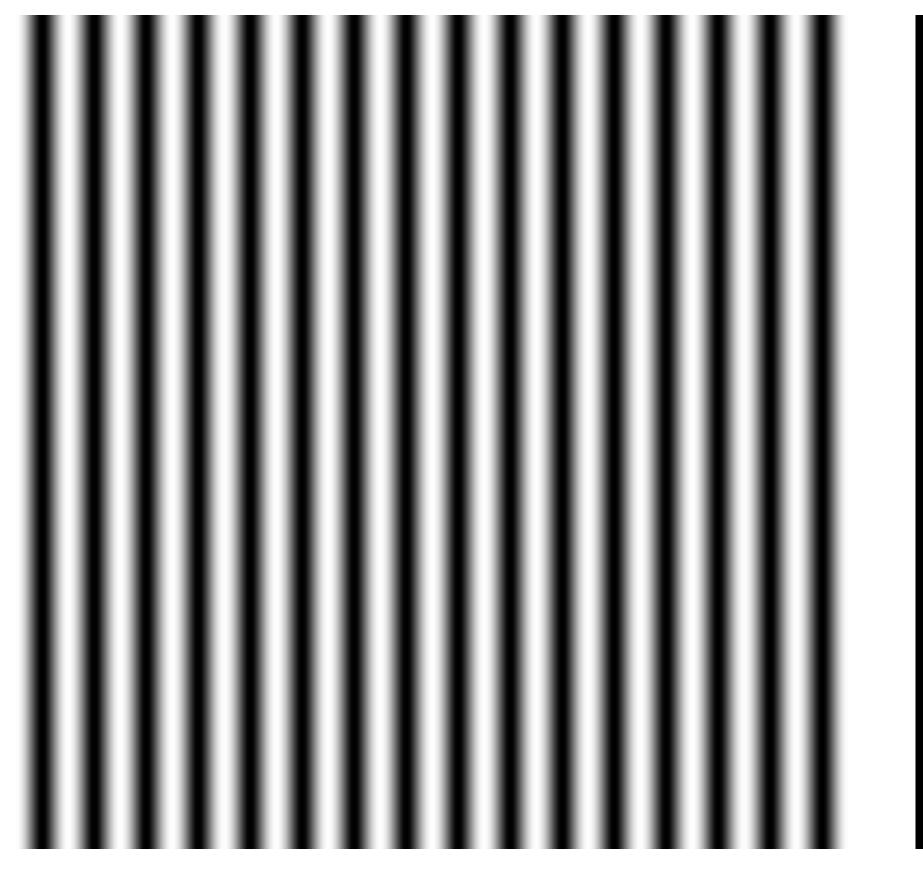


Spatial domain

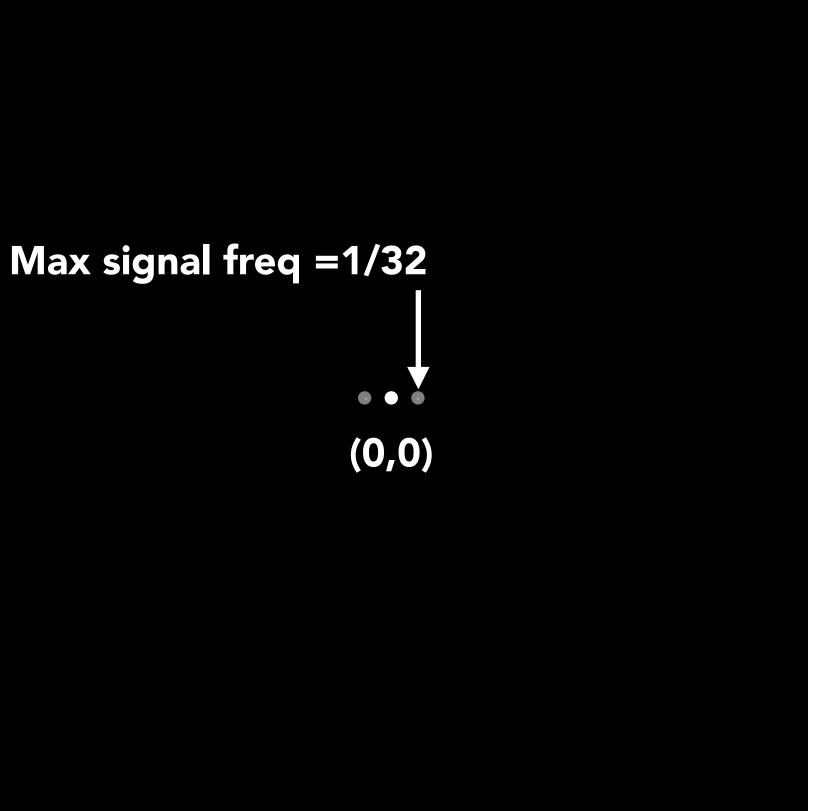


Frequency domain

$\sin(2\pi/32)x$ — frequency 1/32; 32 pixels per cycle

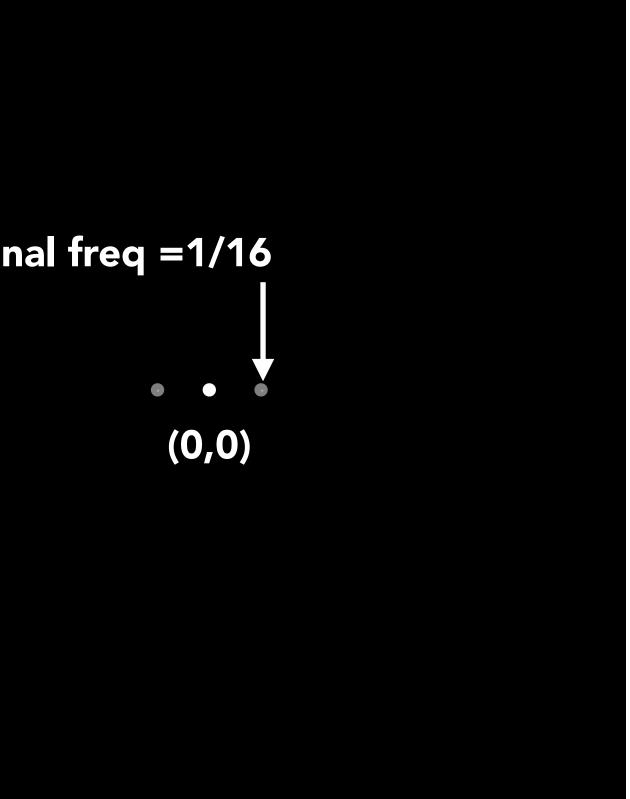


Spatial domain



Frequency domain

$\sin(2\pi/16)x$ — frequency 1/16; 16 pixels per cycle Max signal freq =1/16 (0,0) **Frequency domain Spatial domain**

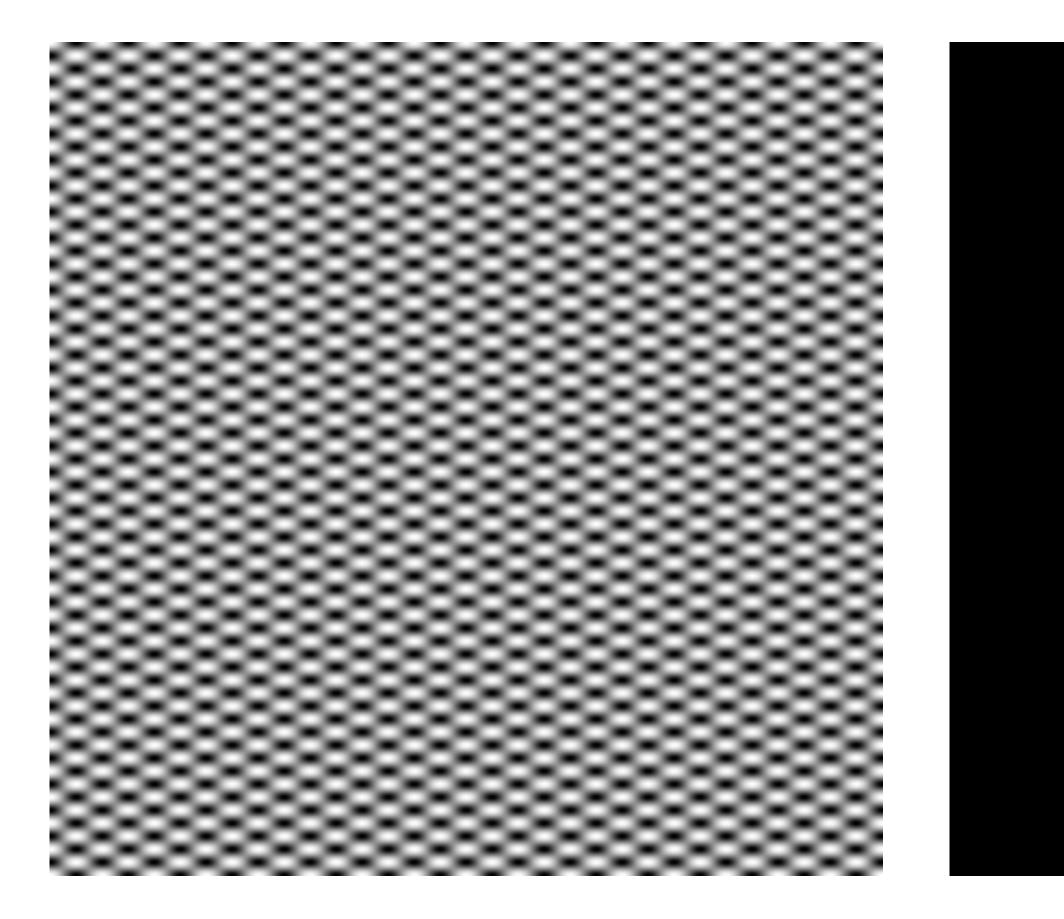


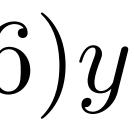
$\sin(2\pi/16)y$

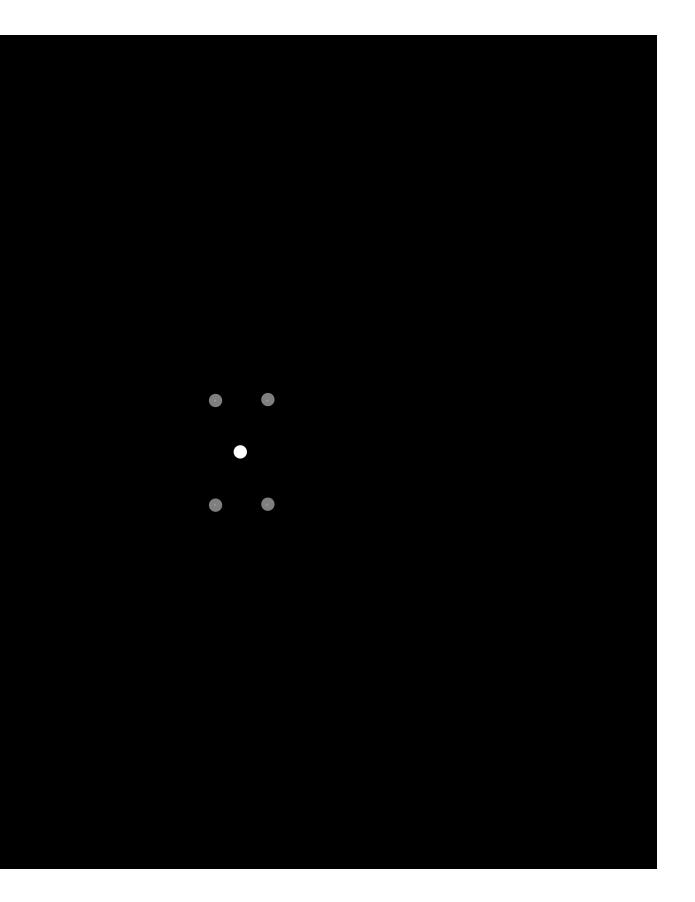
Spatial domain

Frequency domain

 $\sin(2\pi/32)x \times \sin(2\pi/16)y$



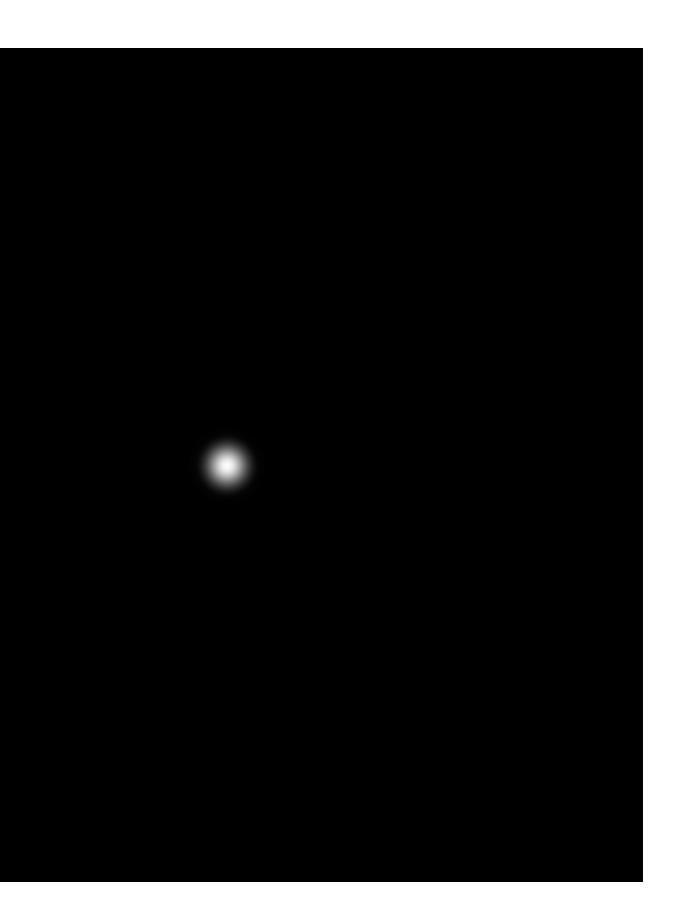




Frequency domain

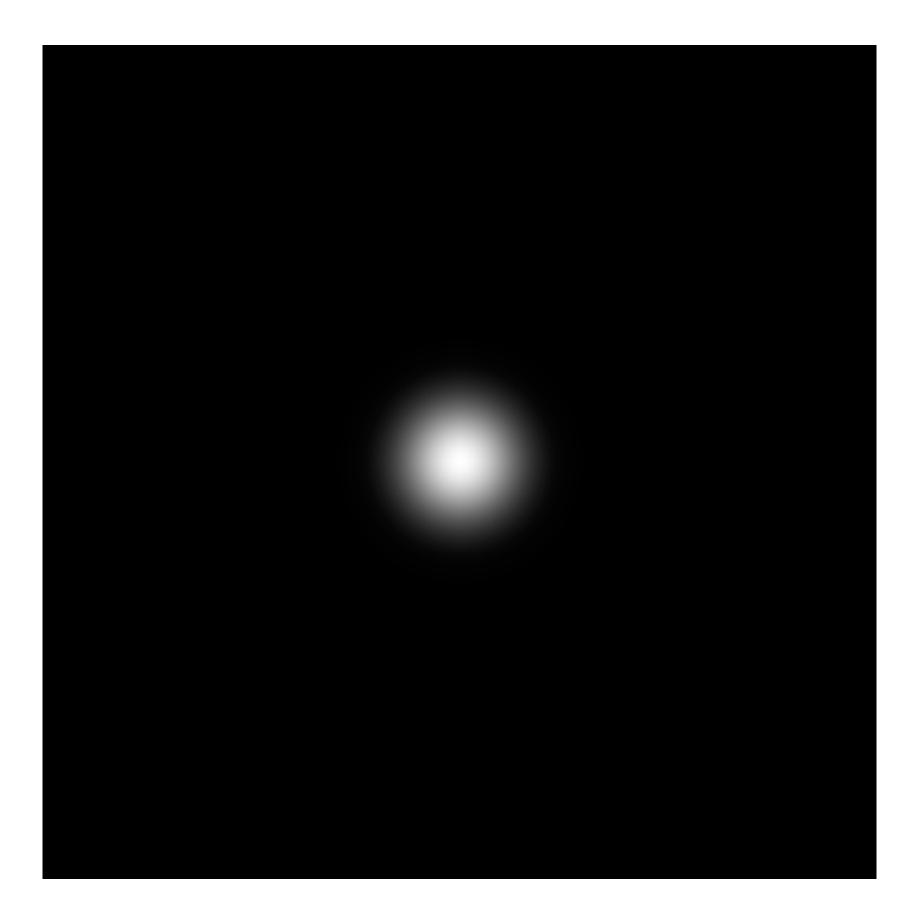
 $\exp(-r^2/16^2)$

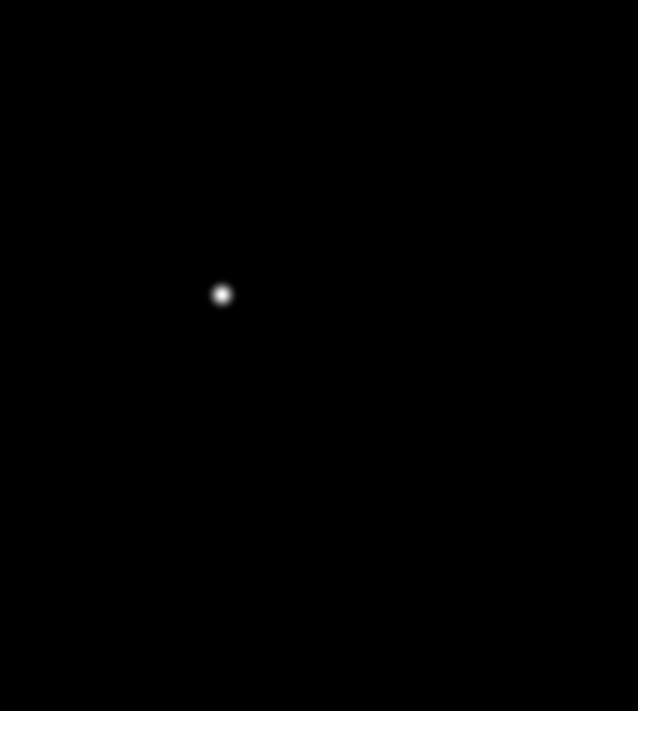




Frequency domain

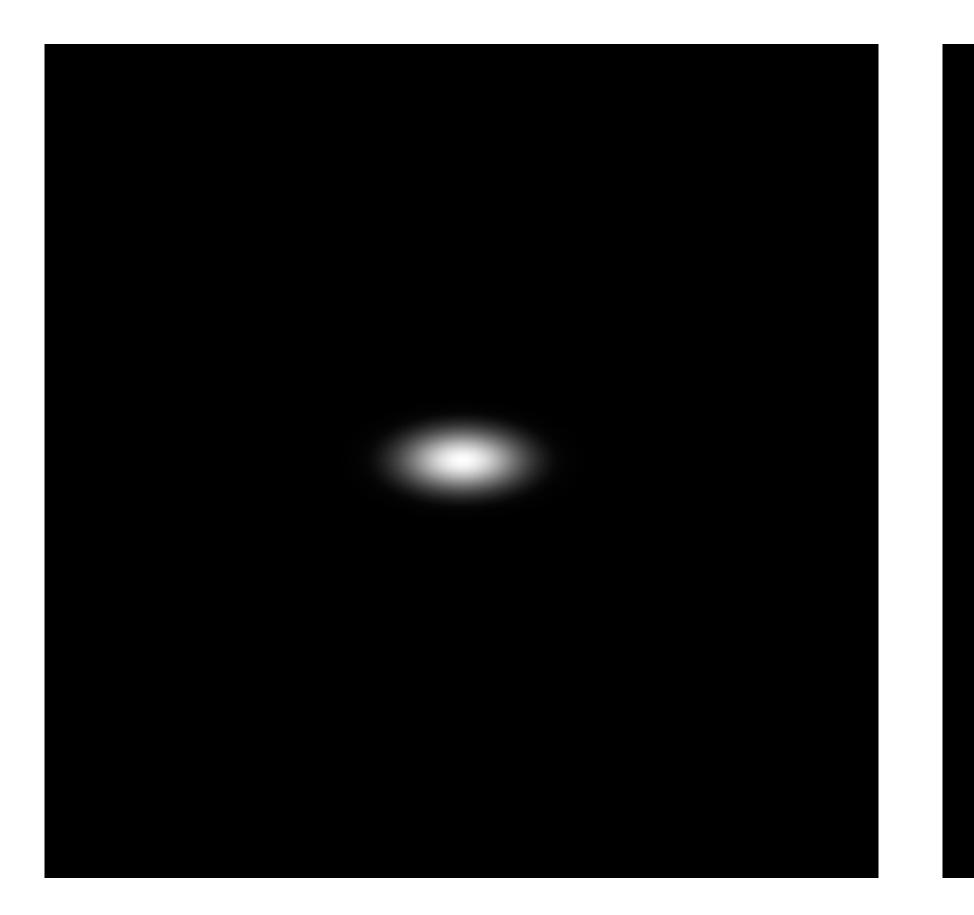
 $\exp(-r^2/32^2)$

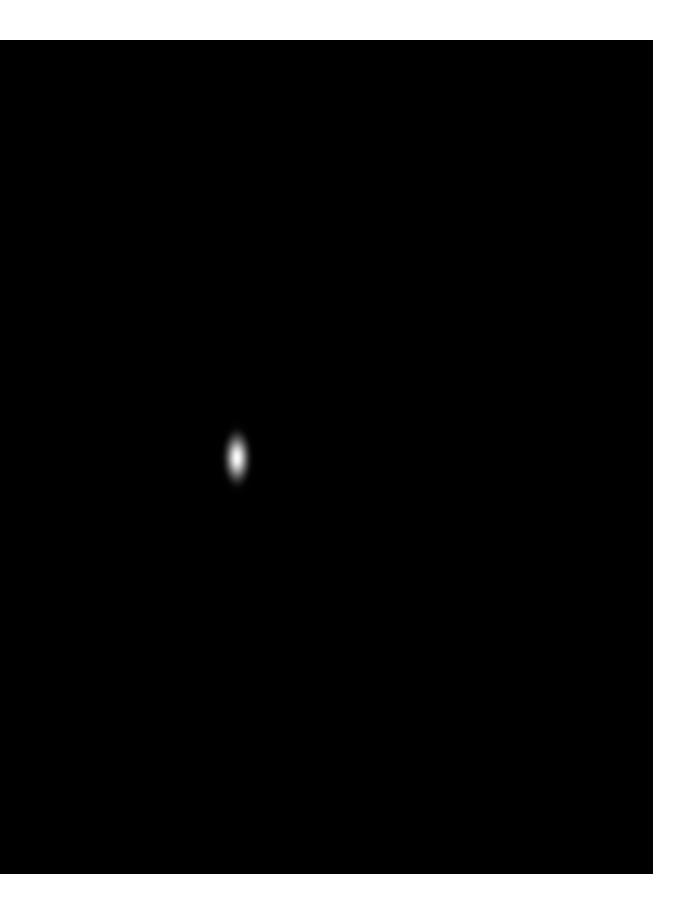




Frequency domain

 $\exp(-x^2/32^2) \times \exp(-y^2/16^2)$





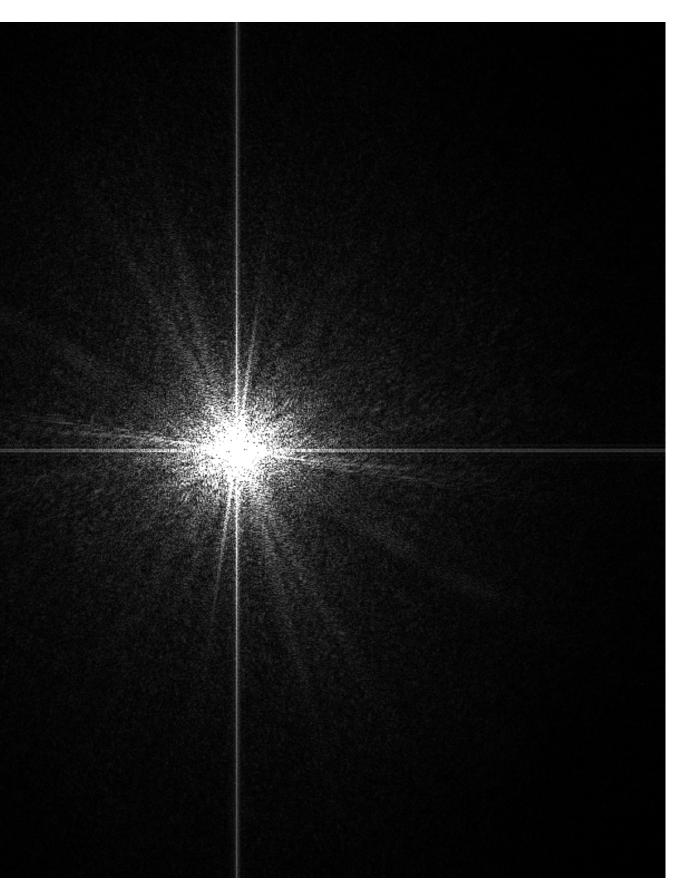
Frequency domain

Image filtering (in the frequency domain)

Manipulating the frequency content of images



Spatial domain



Frequency domain

Low frequencies only (smooth gradients)



Spatial domain

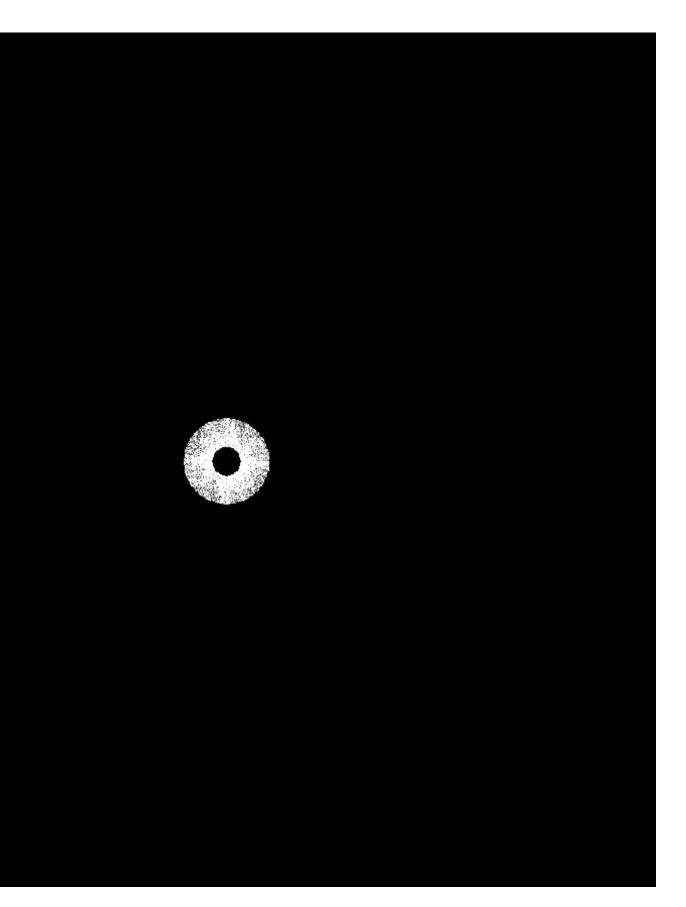
(after low-pass filter) All frequencies above cutoff have 0 magnitude

Frequency domain

Mid-range frequencies

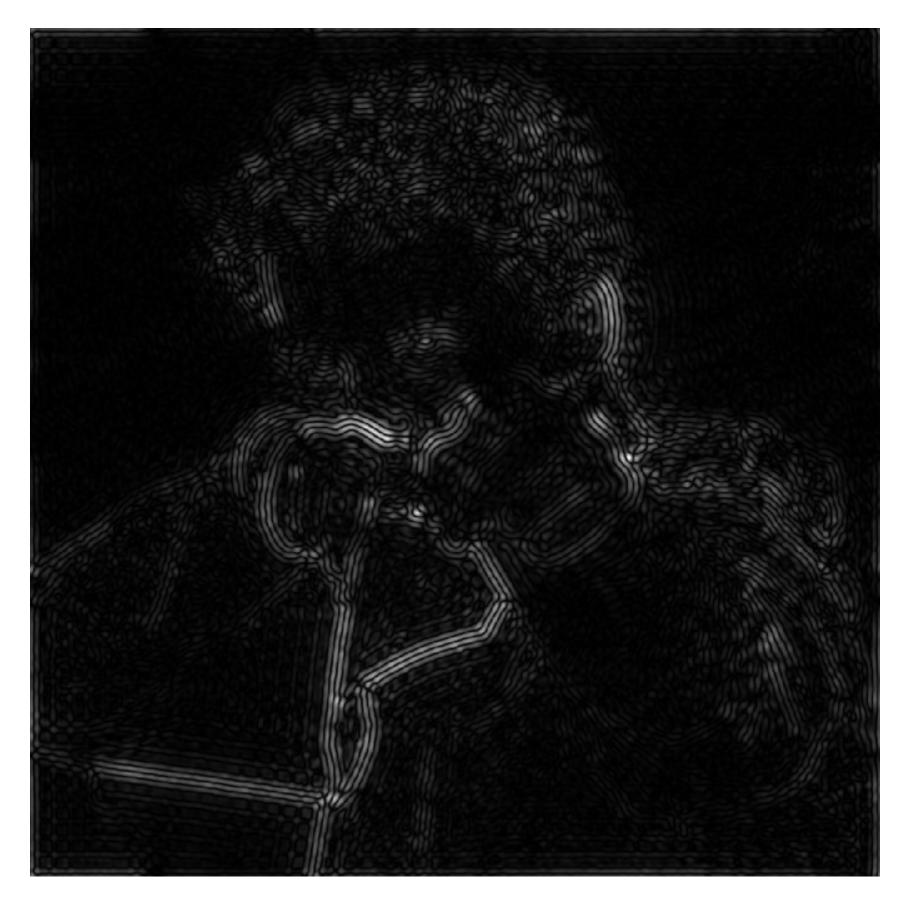


Spatial domain

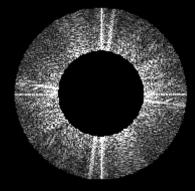


Frequency domain (after band-pass filter)

Mid-range frequencies



Spatial domain

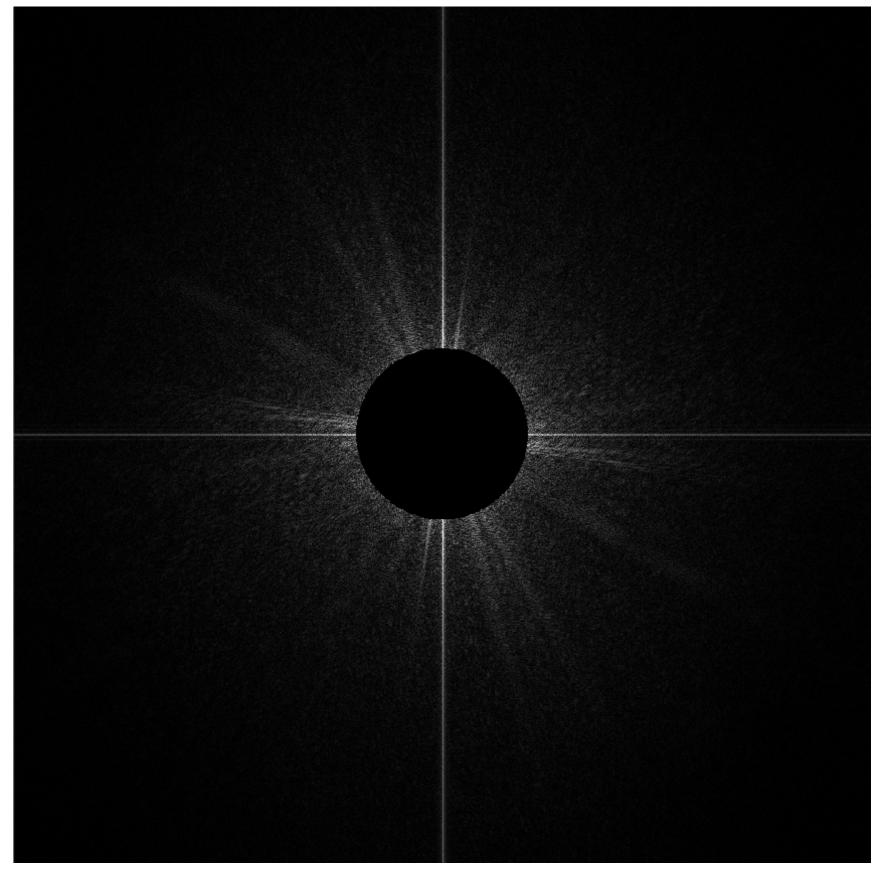


Frequency domain (after band-pass filter)

High frequencies (edges)



Spatial domain (strongest edges)

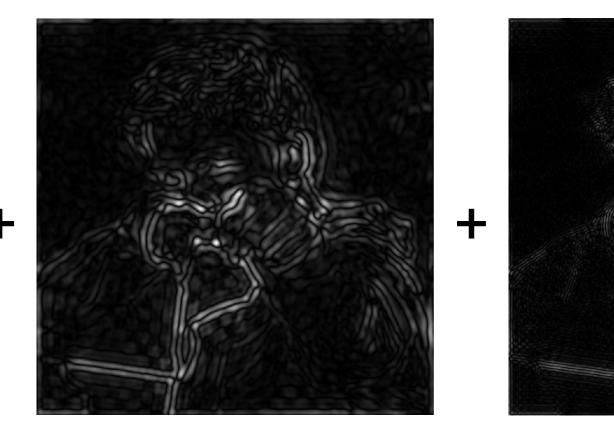


Frequency domain

(after high-pass filter) All frequencies below threshold have 0 magnitude

An image as a sum of its frequency components





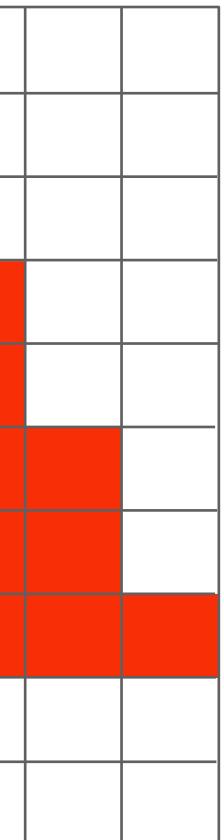




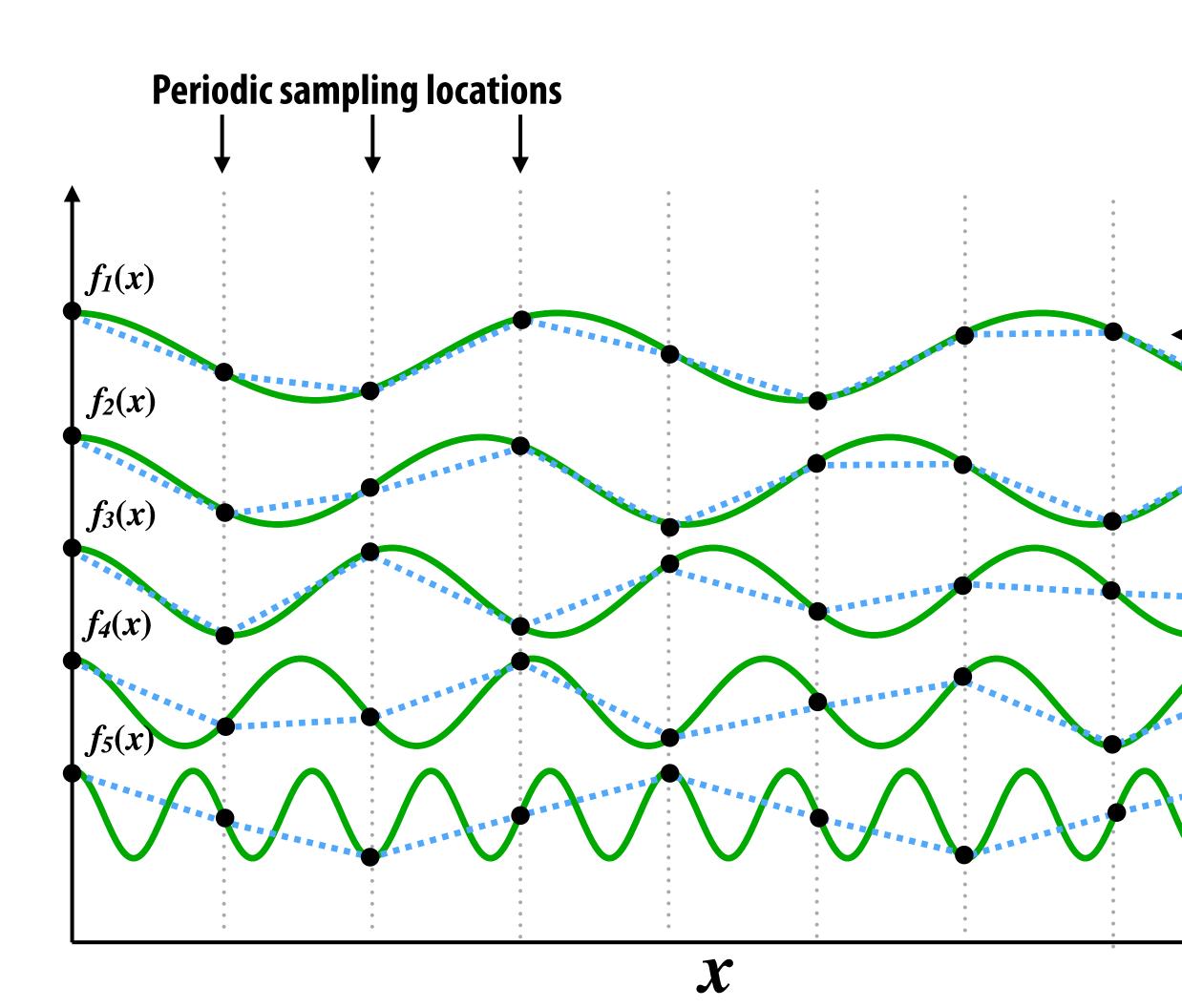


Back to our problem of artifacts in images





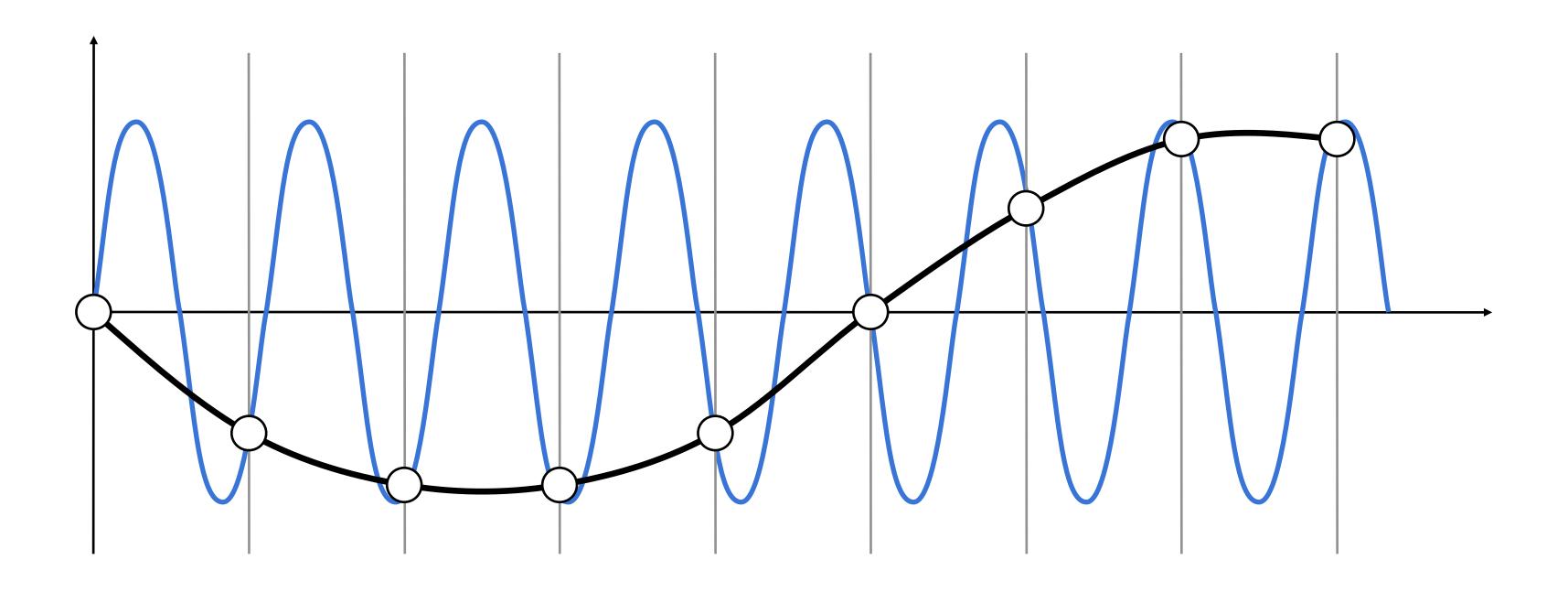
Higher frequencies need denser sampling





High-frequency signal is insufficiently sampled: reconstruction incorrectly appears to be from a low frequency signal

Undersampling creates frequency "aliases"

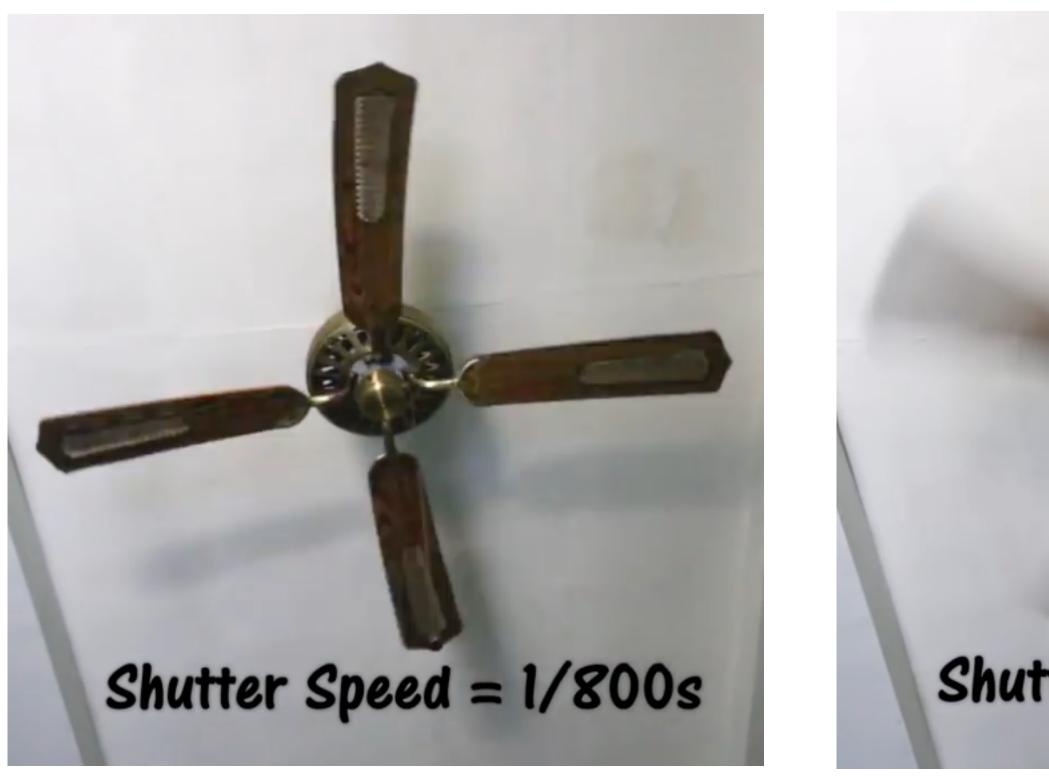


High-frequency signal is insufficiently sampled: samples erroneously appear to be from a low-frequency signal

Two frequencies that are indistinguishable at a given sampling rate are called "aliases"

Anti-aliasing idea: filter out high frequencies before sampling

Video: point vs antialiased sampling



Point in time

Shutter Speed = 1/30s

Motion blurred

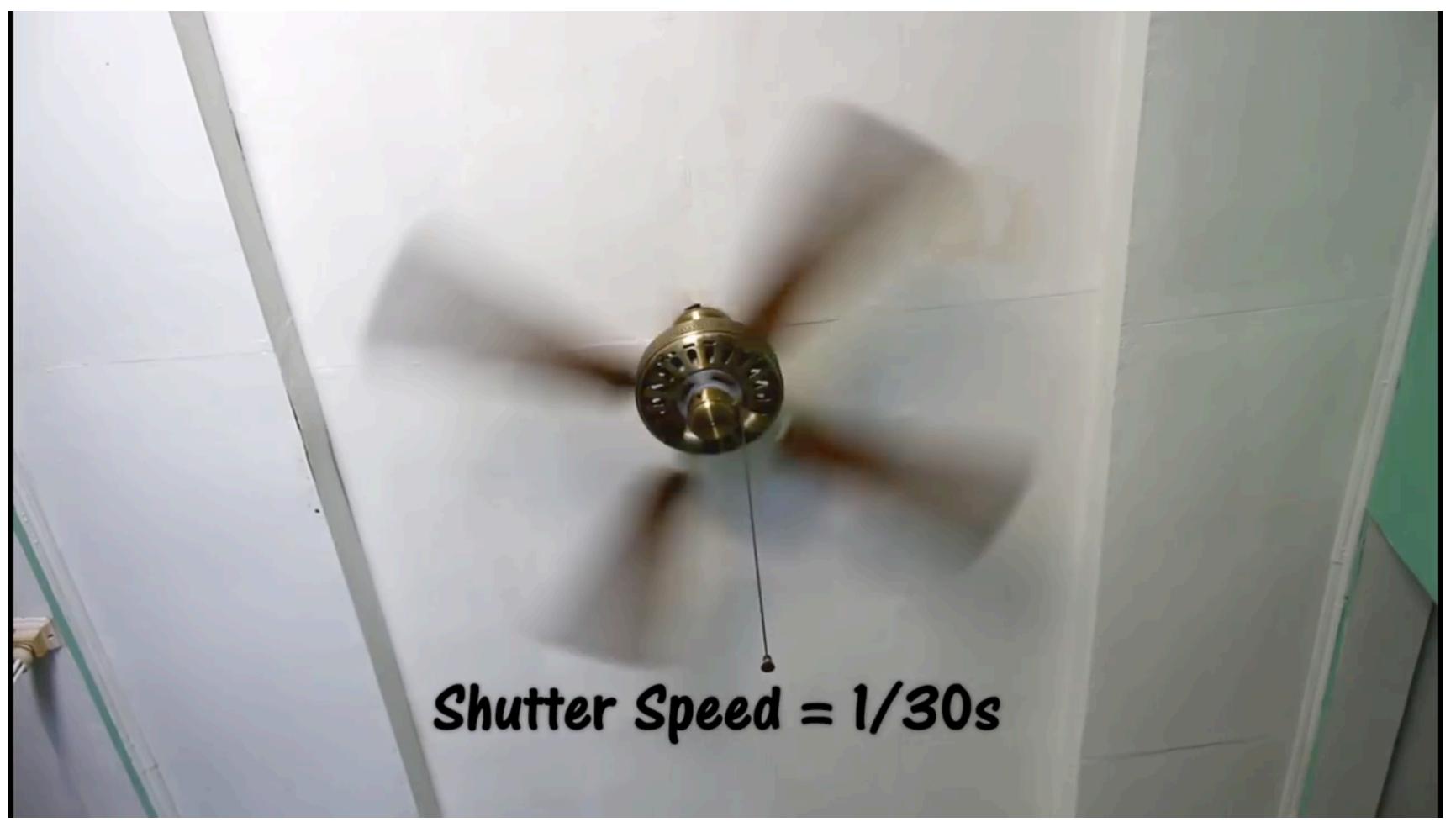
Video: point sampling in time



30 fps video. 1/800 second exposure is sharp in time, causes time aliasing.



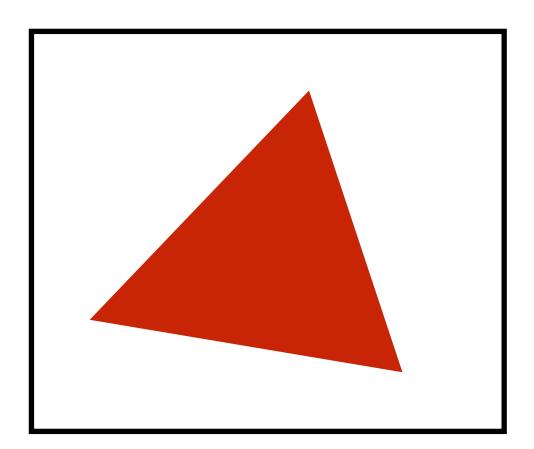
Video: motion-blurred sampling



30 fps video. 1/30 second exposure is motion-blurred in time, reduces aliasing.

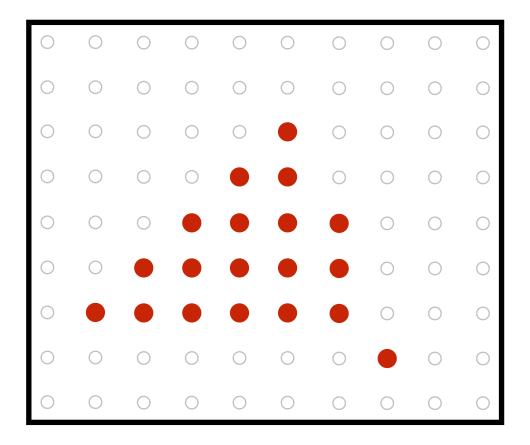


Rasterization is sampling in 2D space

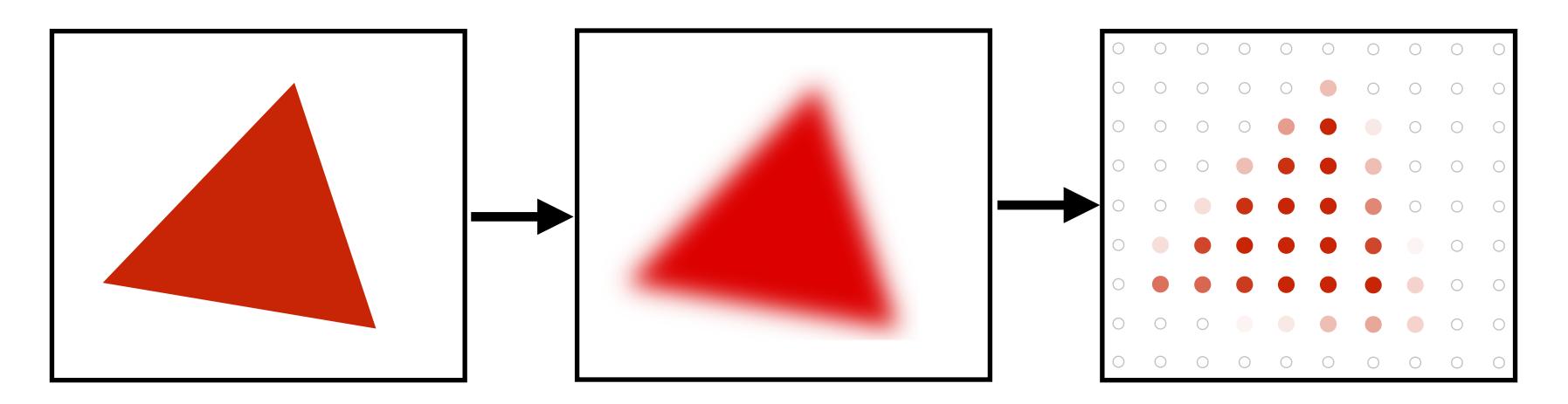


Sample

Note jaggies in rasterized triangle (pixel values are either red or white: sample is in or out of triangle)



Anti-aliasing by pre-filtering the signal



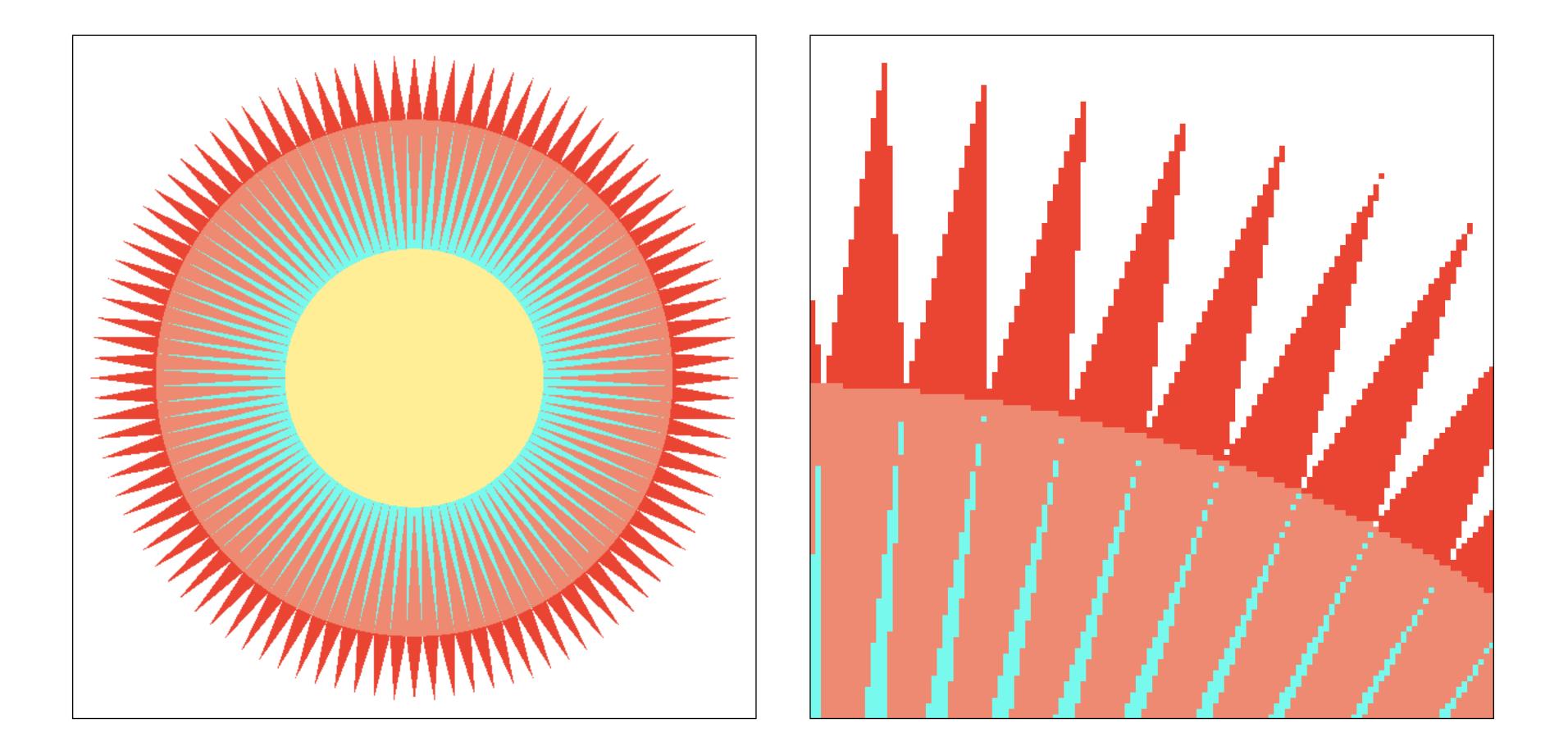


(remove frequencies above Nyquist limit)

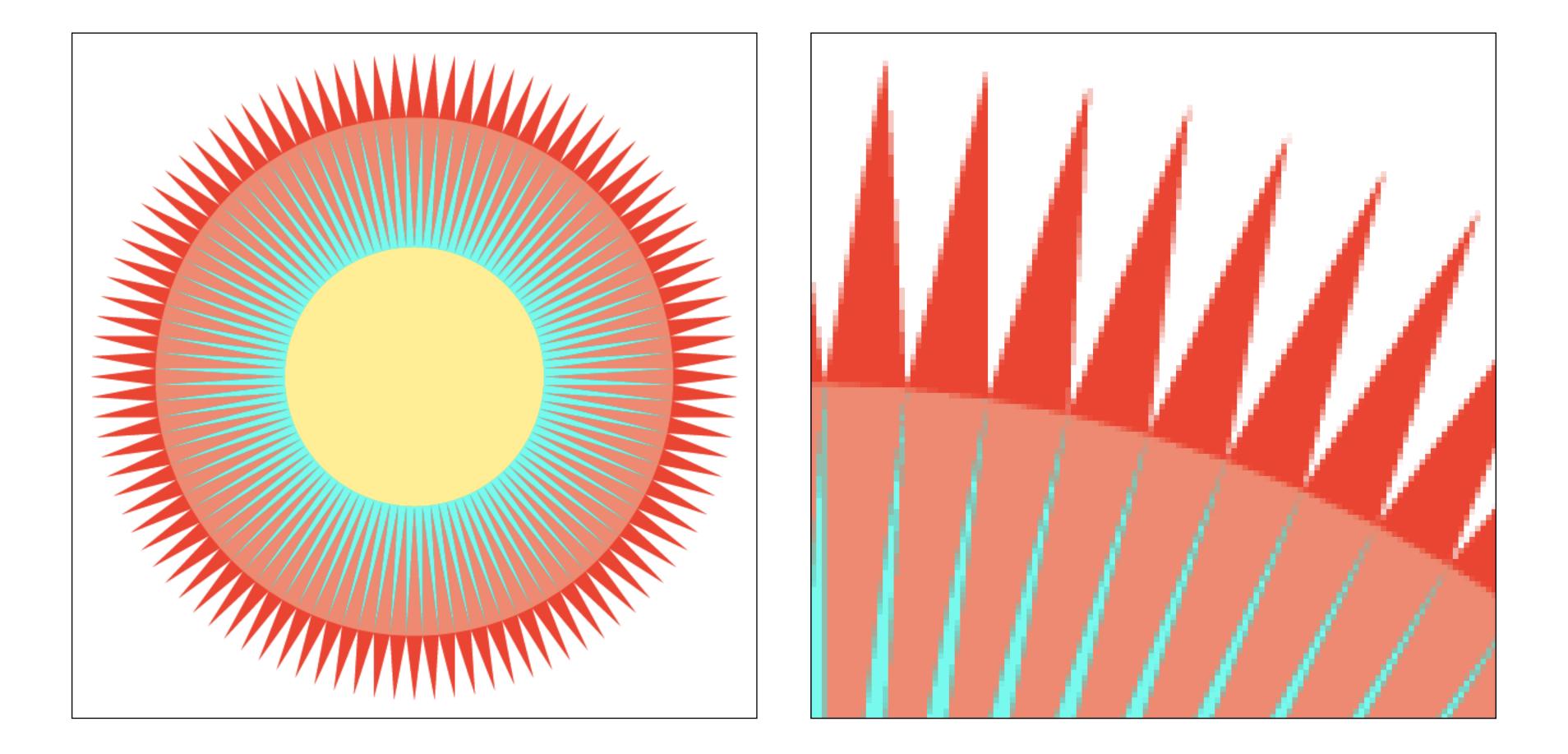
Note anti-aliased edges of rasterized triangle: pixel values take intermediate values

Sample

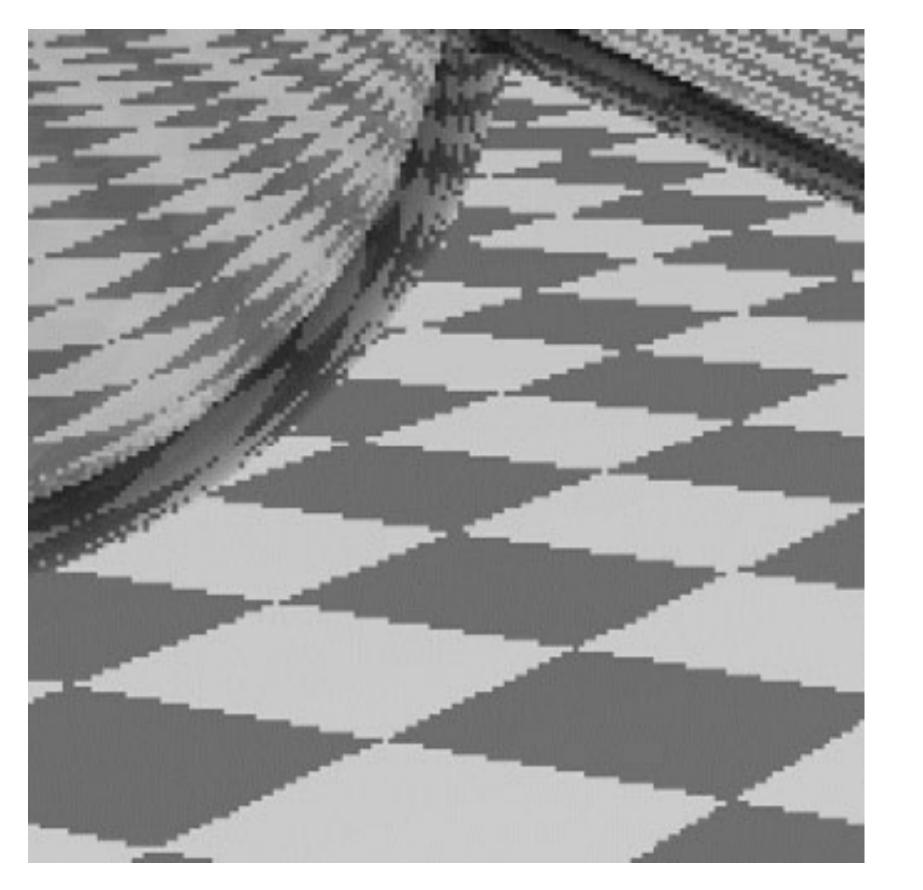
Images rendered using one sample per pixel

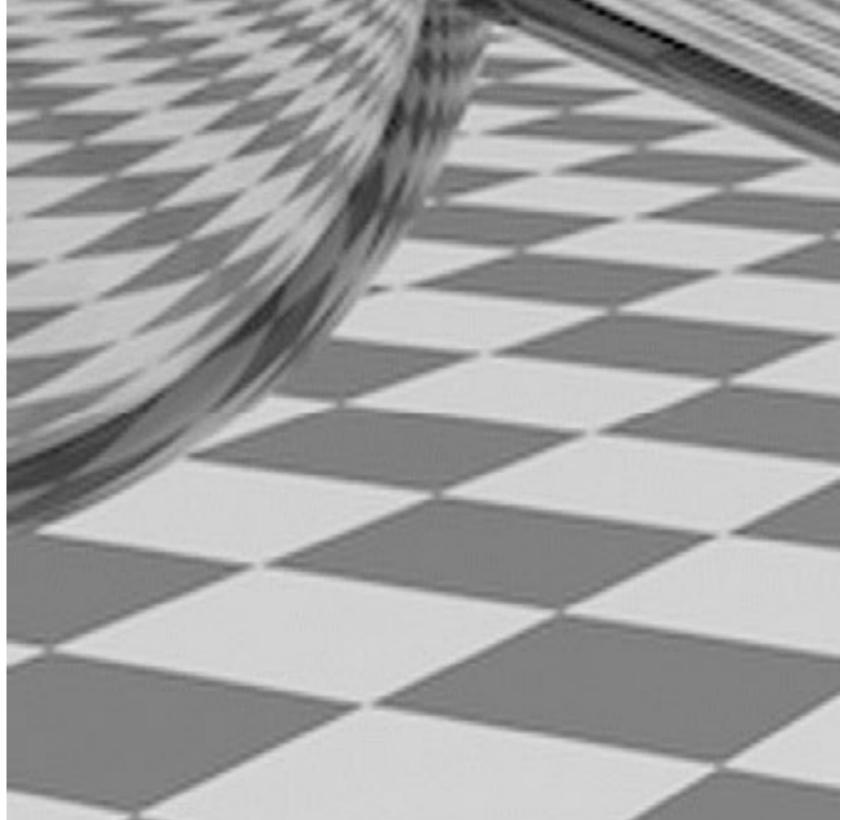


Anti-aliased results



Benefits of anti-aliasing

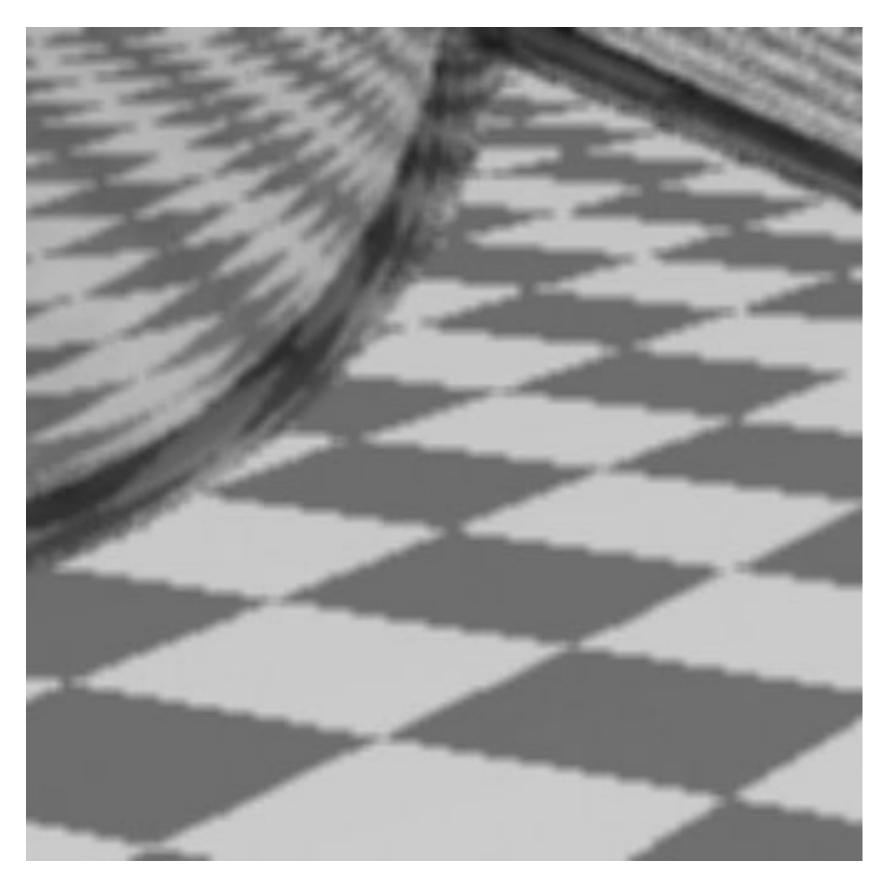


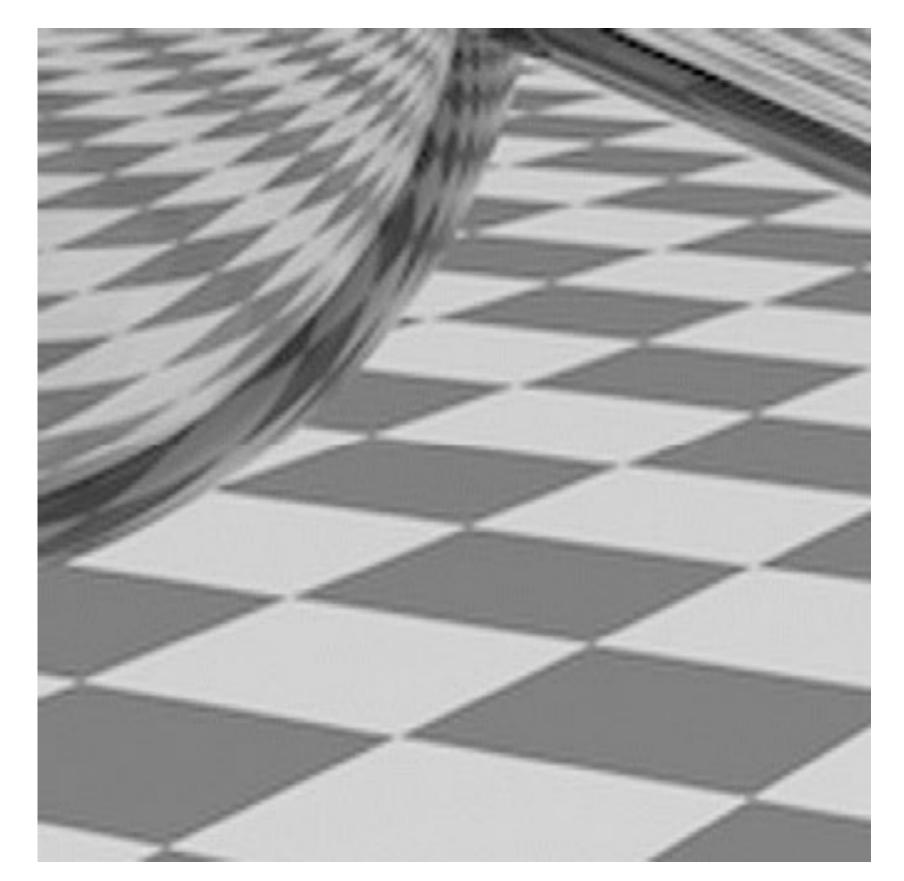


Jaggies

Pre-filtered

Anti-aliasing vs blurring an aliased result

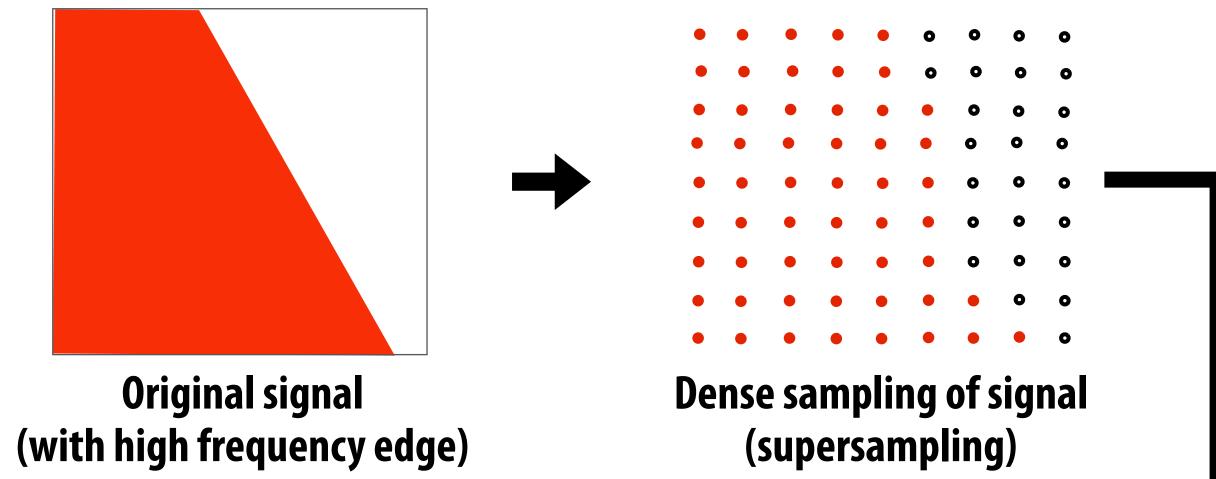


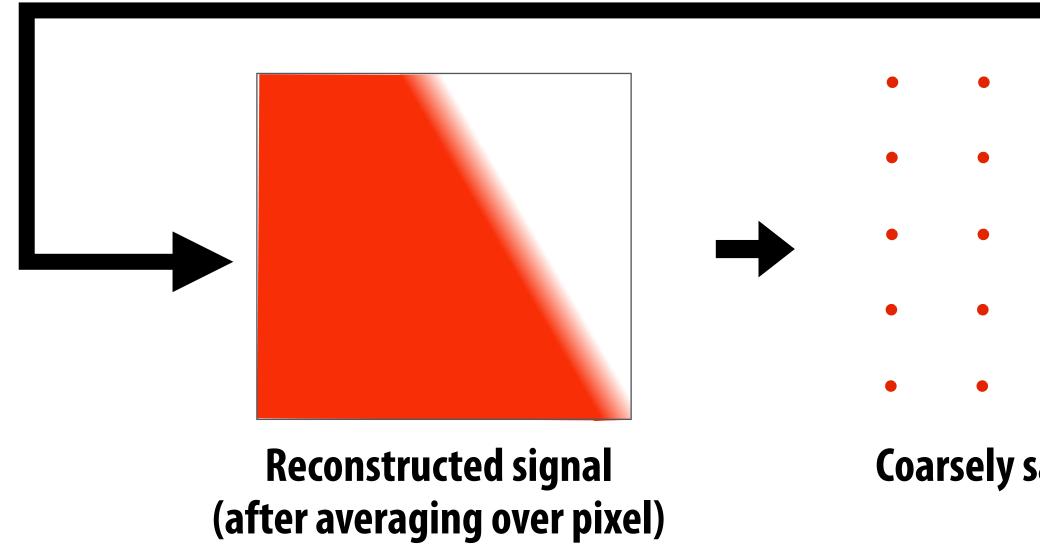


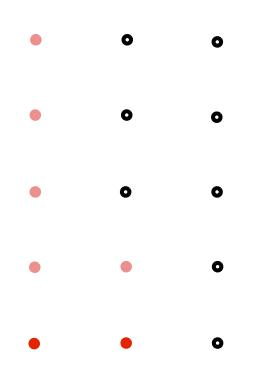
Blurred Jaggies ("Sample then blur jaggies")

Pre-filtered ("blur then sample")

Recall our anti-aliasing technique from the first half of lecture

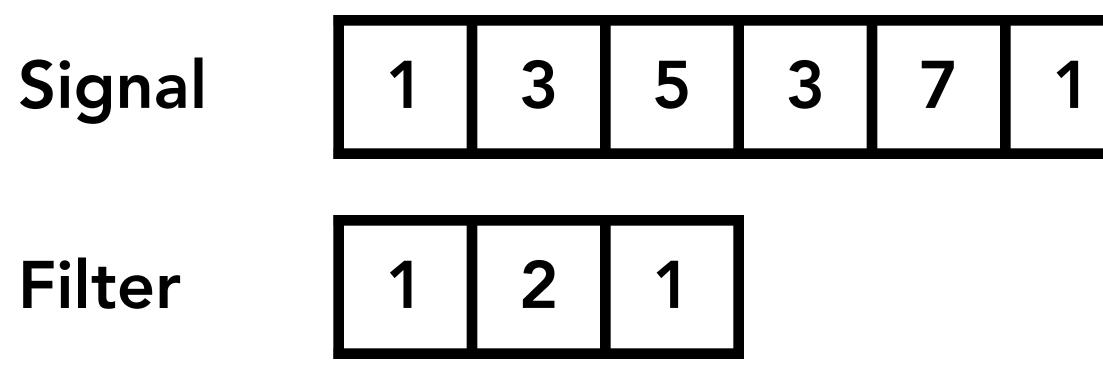




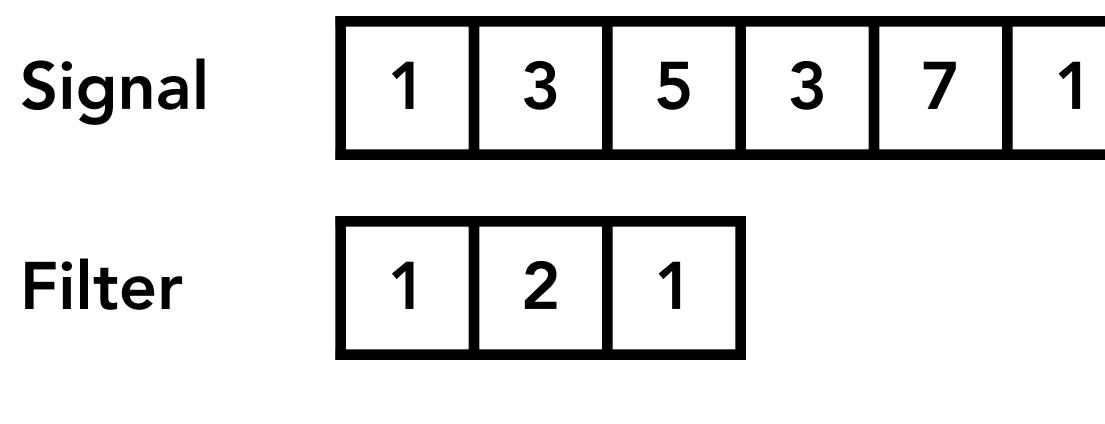


Coarsely sampled signal

Filtering = convolution

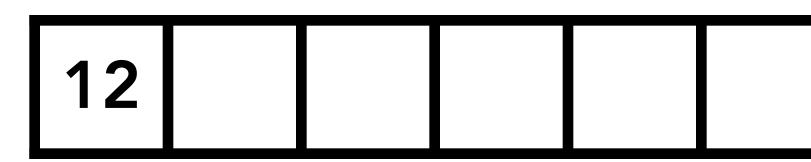


3 8	6	4
-----	---	---

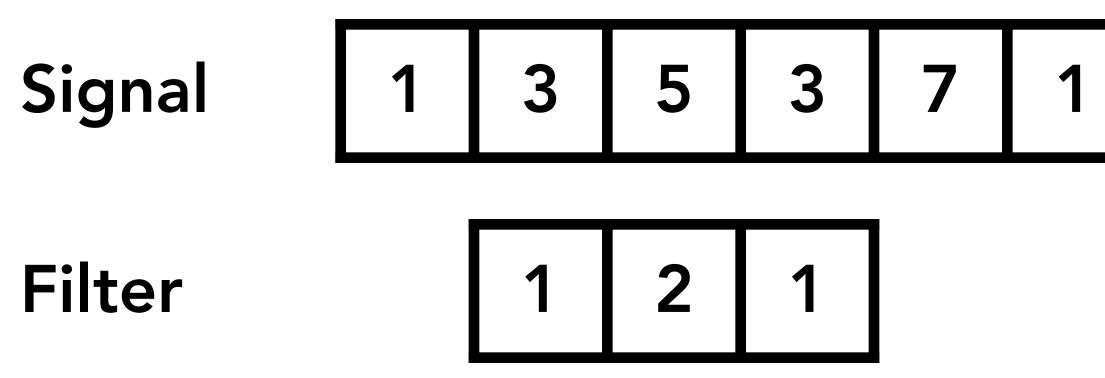


1x1 + 3x2 + 5x1 = 12

Result



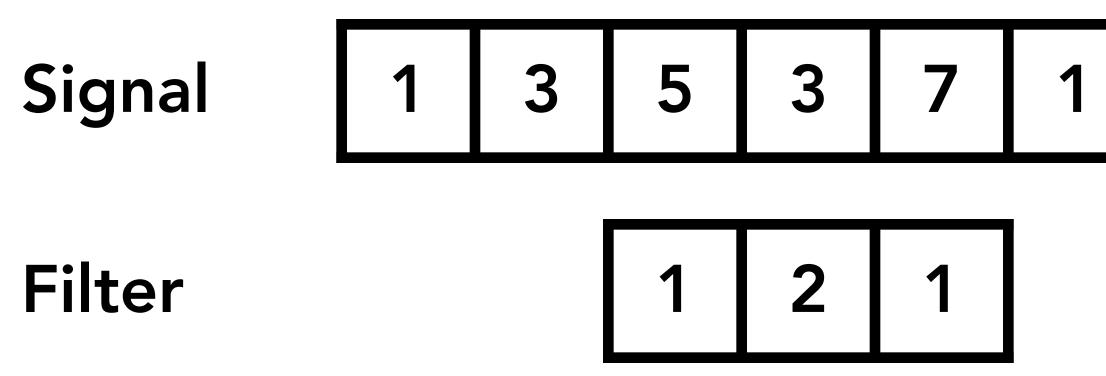
3 8	6	4
-----	---	---



3x1 + 5x2 + 3x1 = 16

Result

3 8	6	4
-----	---	---

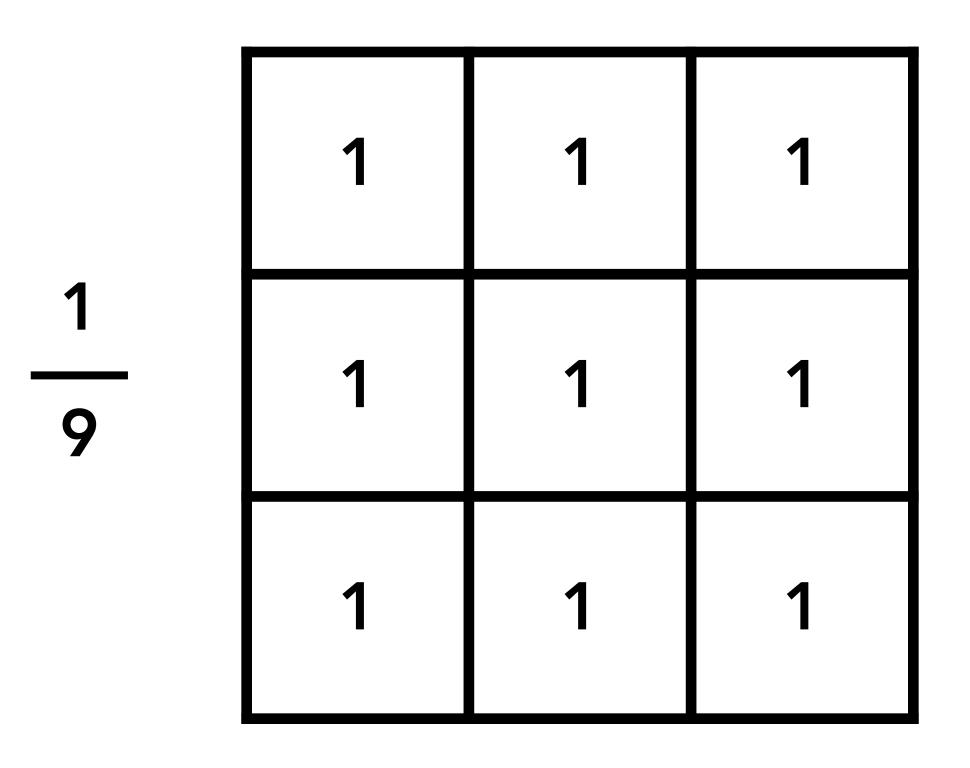


5x1 + 3x2 + 7x1 = 18

Result

3 8	6	4
-----	---	---

Box filter (used in a 2D convolution)



Example: 3x3 box filter

2D convolution with box filter blurs the image

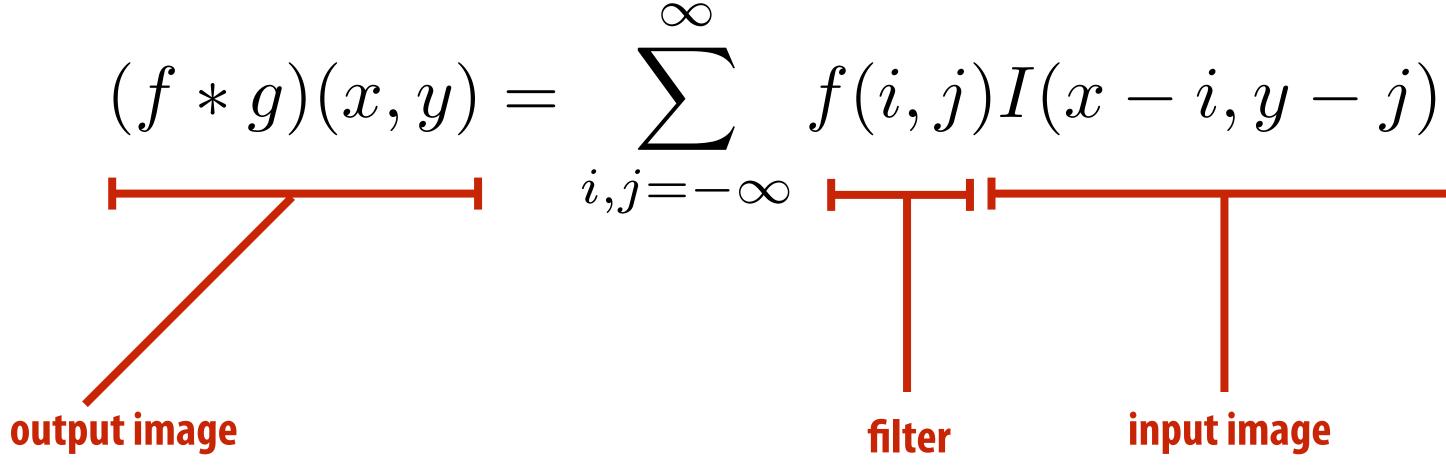


Original image

Hmm... this reminds me of a low-pass filter...

Blurred (convolve with box filter)

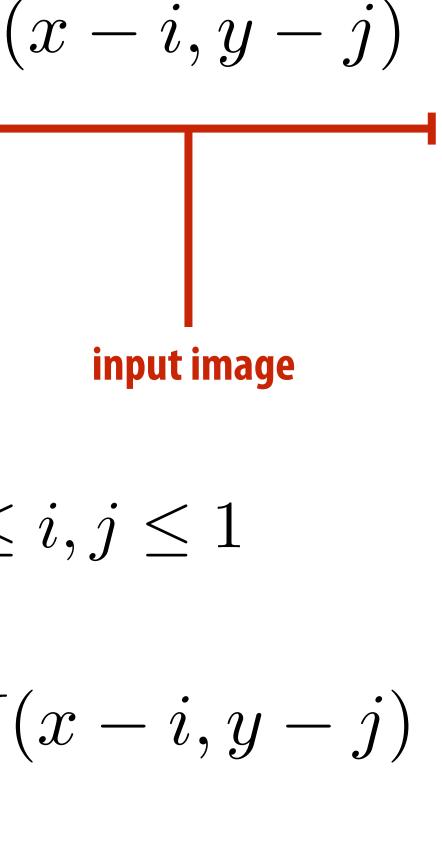
Discrete 2D convolution



Consider f(i, j) that is nonzero only when: $-1 \le i, j \le 1$ Then: $(f * g)(x, y) = \sum f(i, j)I(x - i, y - j)$ i, j = -1

And we can represent f(i,j) as a 3x3 matrix of values where:

$$f(i,j) = \mathbf{F}_{i,j}$$
 (often called: "fi



ilter weights", "filter kernel")

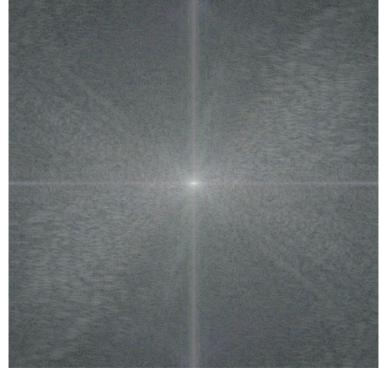
Convolution theorem Convolution in the spatial domain is equal to multiplication in the frequency domain, and vice versa

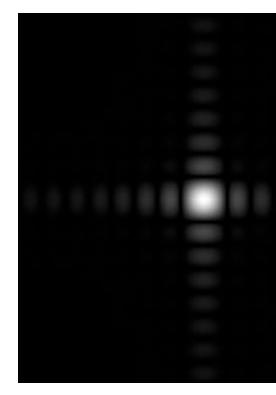
Spatial Domain

Frequency Domain

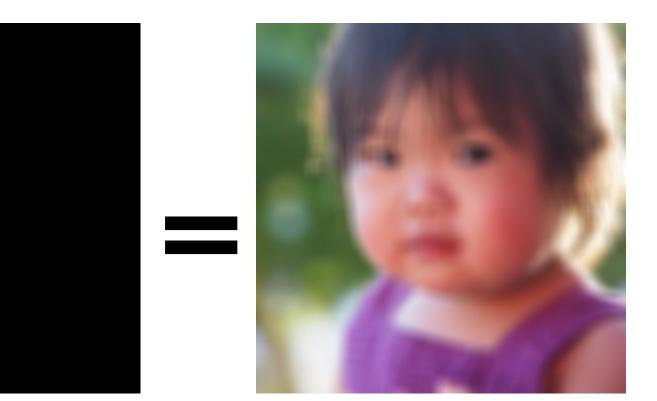


Fourier Transform

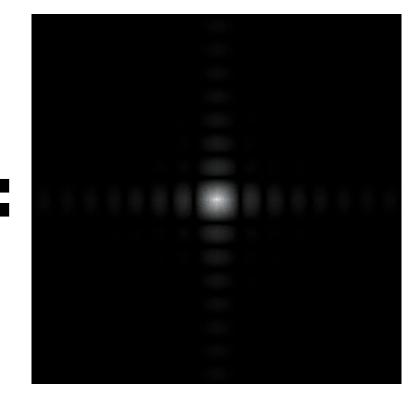




X



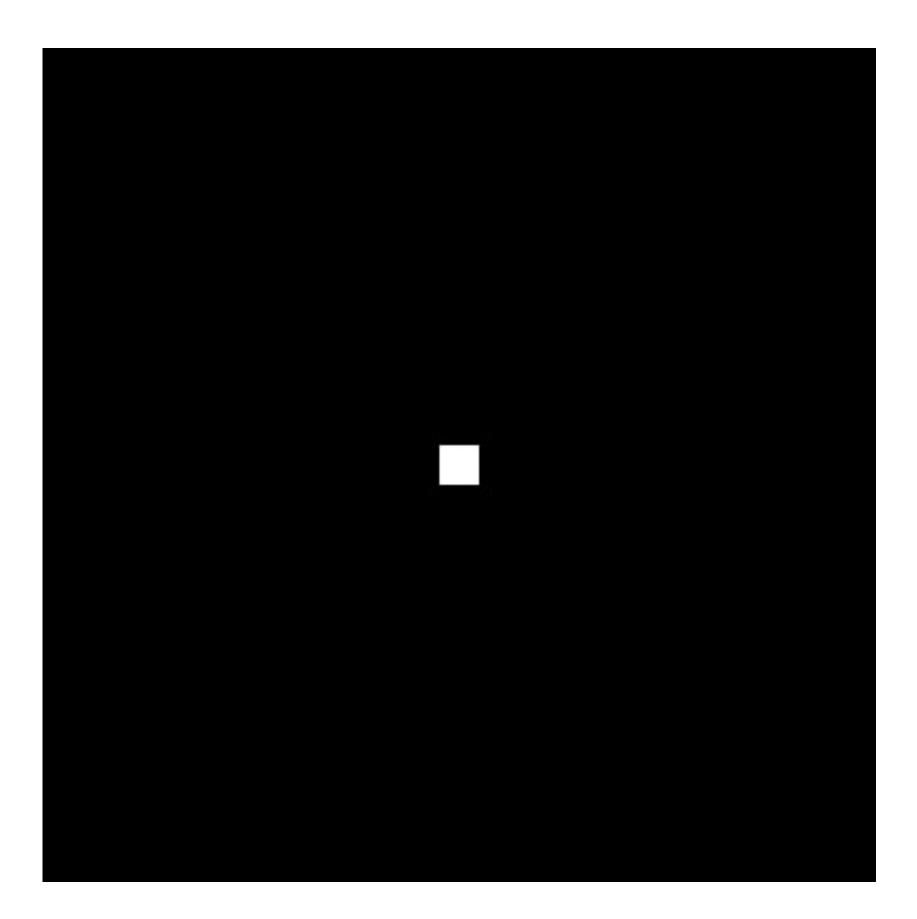
Inv. Fourier Transform



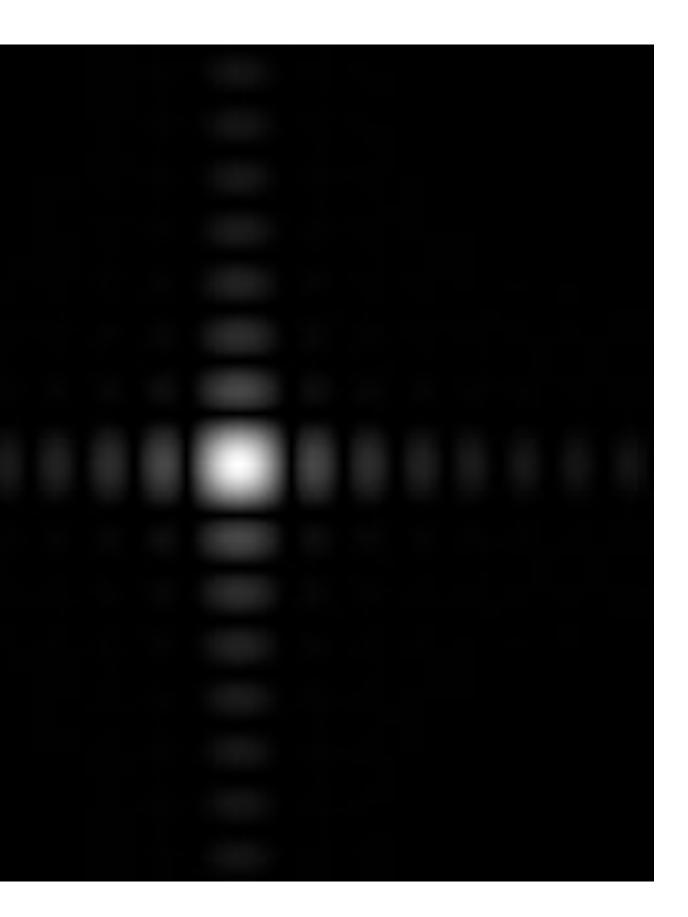
Convolution theorem

- Convolution in the spatial domain is equal to multiplication in the frequency domain, and vice versa
- Pre-filtering option 1:
 - Filter by convolution in the spatial domain
- Pre-filtering option 2:
 - Transform to frequency domain (Fourier transform)
 - Multiply by Fourier transform of convolution kernel
 - Transform back to spatial domain (inverse Fourier)

Box function = "low pass" filter

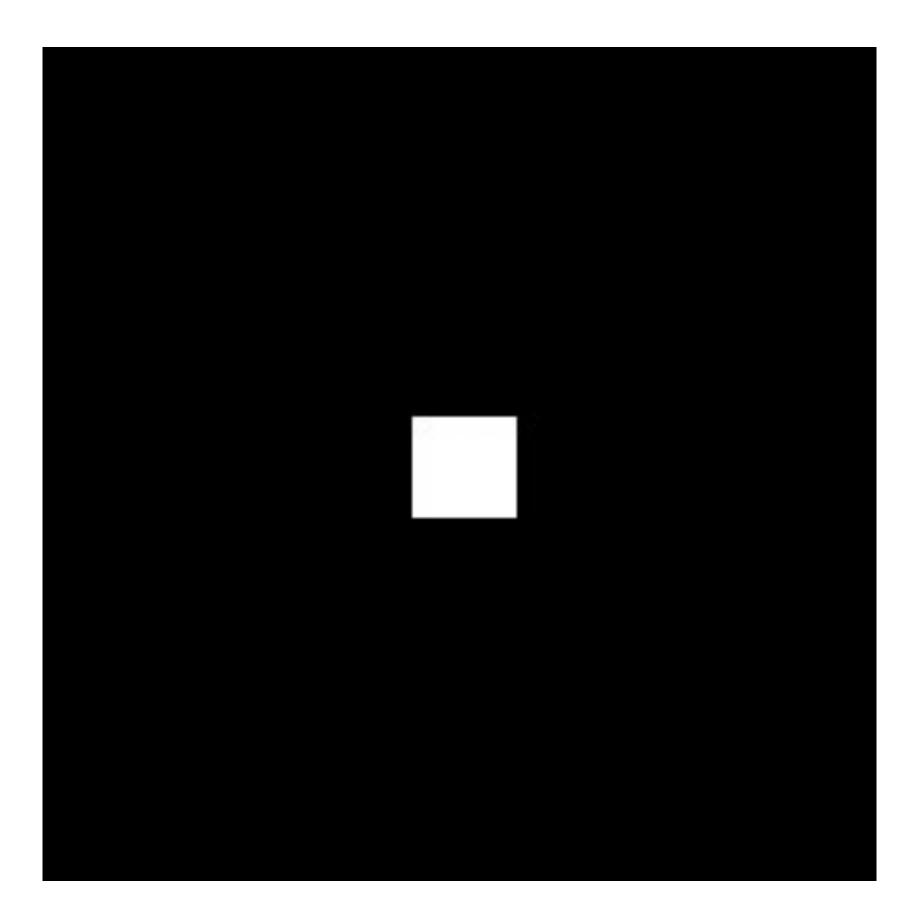






Frequency domain

Wider filter kernel = retain only lower frequencies



Spatial domain



Frequency domain

Wider filter kernel = lower frequencies

- As a filter is localized in the spatial domain, it spreads out in frequency domain
- Conversely, as a filter is localized in frequency domain, it spreads out in the spatial domain

How can we reduce aliasing error?

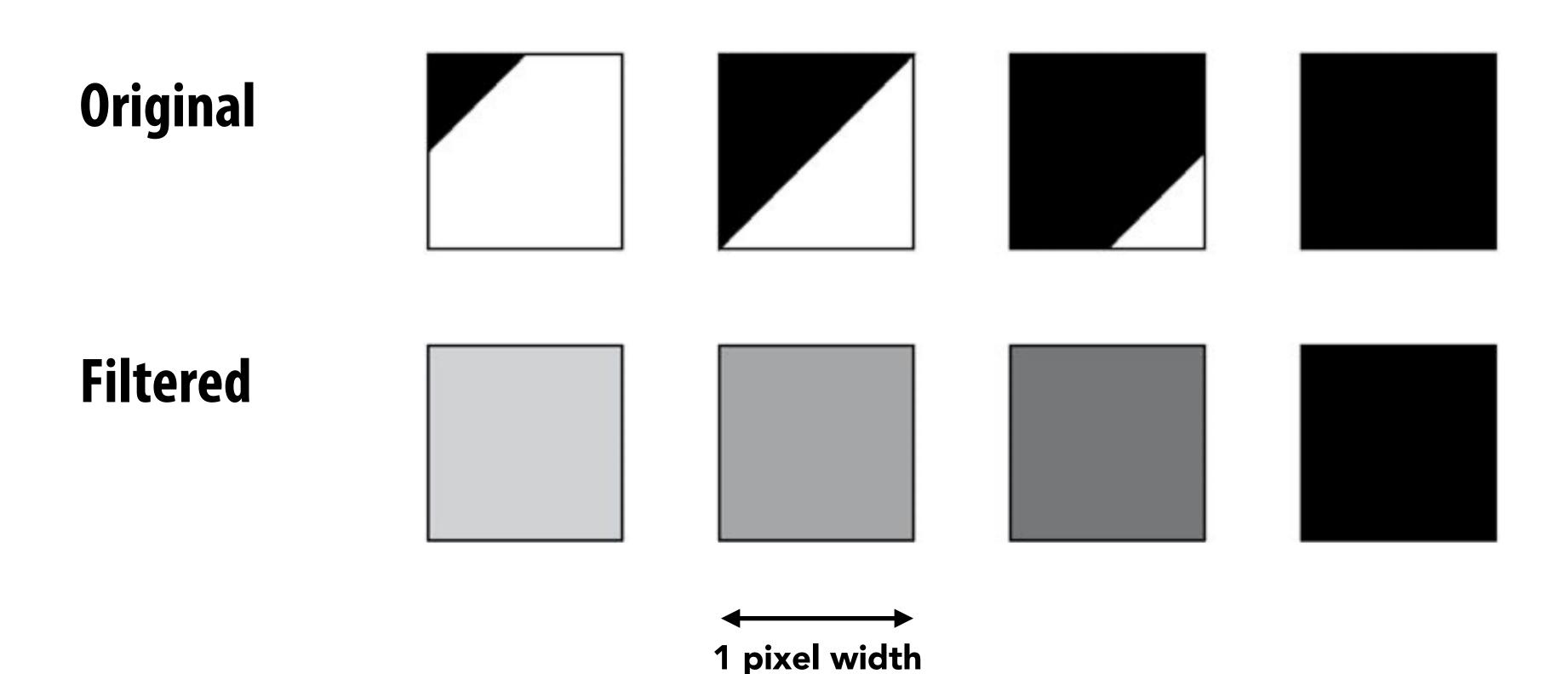
- Increase sampling rate
 - Higher resolution displays, sensors, framebuffers...
 - But: costly and may need very high resolution to sufficiently reduce aliasing
- Anti-aliasing
 - Simple idea: remove (or reduce) high frequencies before sampling
 - How to filter out high frequencies before sampling?

Anti-aliasing by averaging values in pixel area

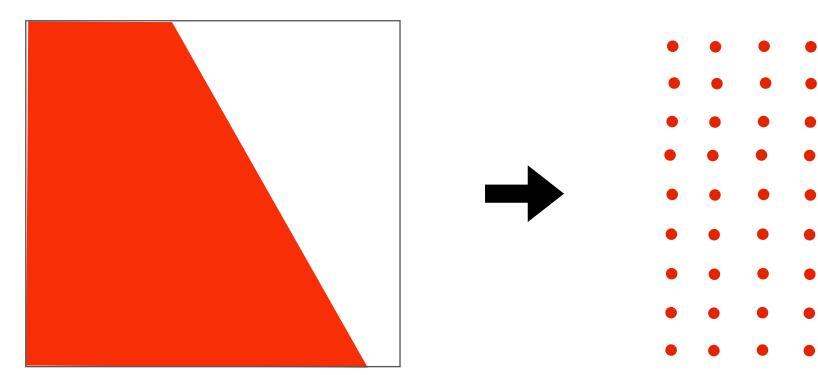
- Convince yourself the following are the same:
- **Option 1:**
 - Convolve f(x,y) by a 1-pixel box-blur
 - Then sample at every pixel
- **Option 2:**
 - Compute the average value of f(x,y) in the pixel

Anti-aliasing by computing average pixel value

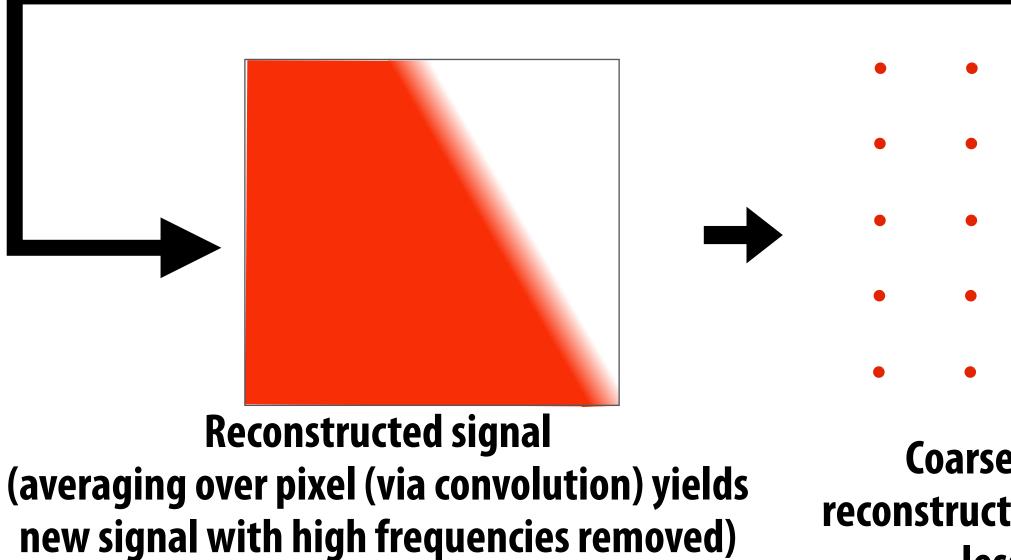
In rasterizing one triangle, the value of f(x,y) = inside(tri,x,y) averaged over the area of a pixel is equal to the amount of the pixel covered by the triangle.



Putting it all together: anti-aliasing via supersampling



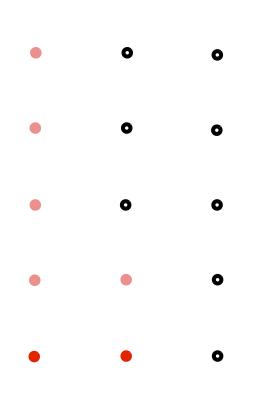
Original signal (with high frequency edge)





•	•	•	•	•	
•	0	0	0	0	
٠	٠	0	0	0	
•	٠	0	0	0	
•	•	0	0	0	
•	•	0	0	0	
•	•	0	0	0	
•	•	•	0	0	
•	•	•	•	0	
				_	

Dense sampling of signal (supersampling)



Coarse sampling of reconstructed signal exhibits less aliasing

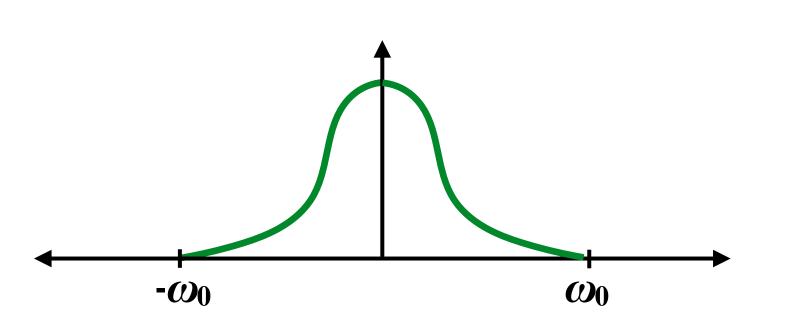
Today's summary

- Drawing a triangle = sampling triangle/screen coverage
- **Pitfall of sampling: aliasing**
- **Reduce aliasing by prefiltering signal**
 - Supersample
 - **Reconstruct via convolution (average coverage over pixel)**
 - Higher frequencies removed
 - Sample reconstructed signal once per pixel
- There is much, much more to sampling theory and practice...

Bonus slides: How much pre-filtering do we need to avoid aliasing?

Nyquist-Shannon theorem

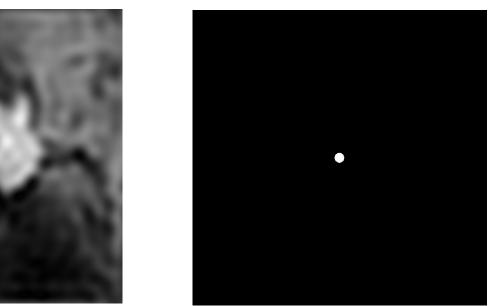
- Consider a band-limited signal: has no frequencies above ω_0
 - **1D: consider low-pass filtered audio signal**
 - 2D: recall the blurred image example from a few slides ago





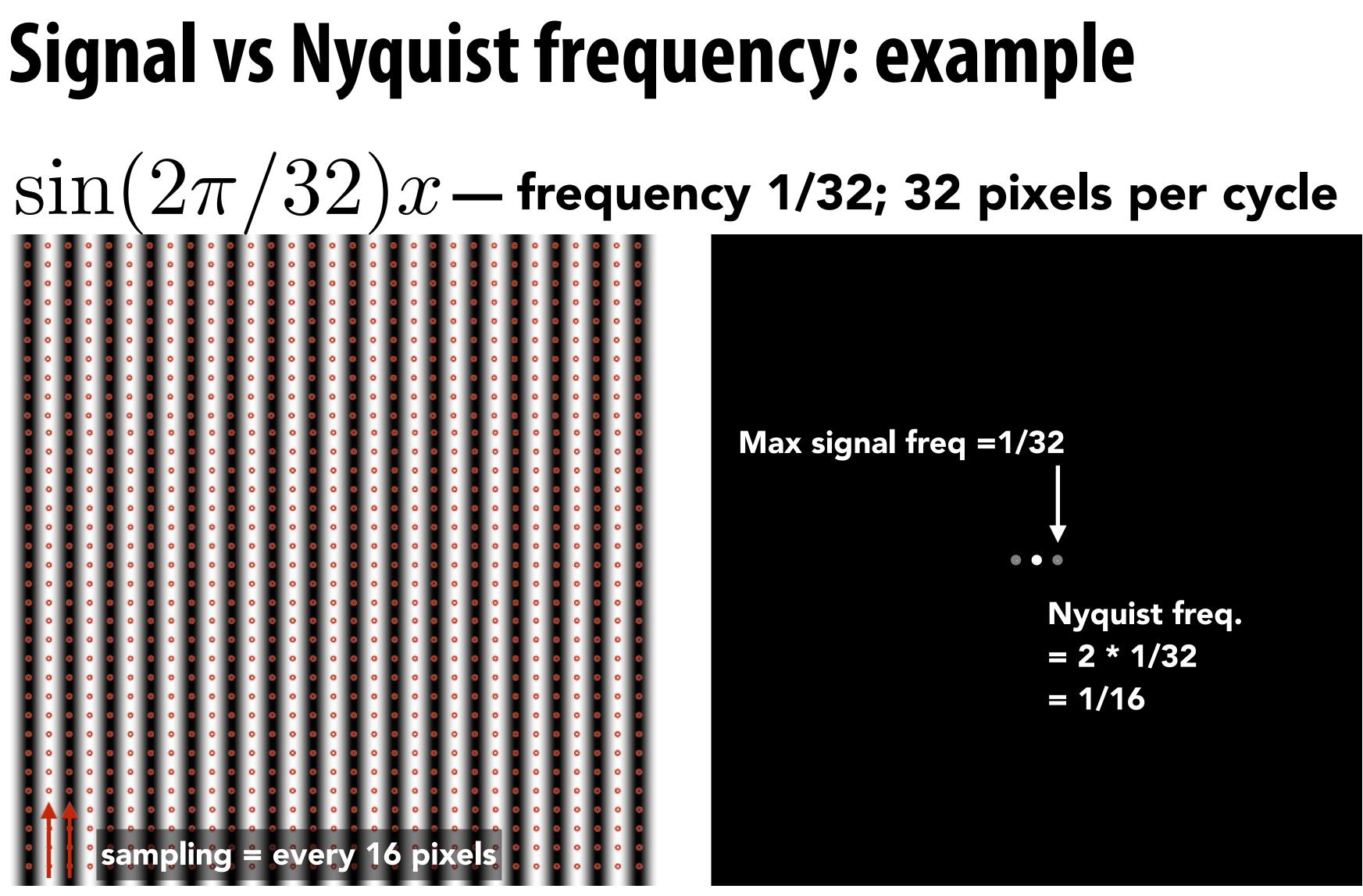
The signal can be perfectly reconstructed if sampled with period $T = 1/2\omega_0$ And reconstruction is performed using a "sinc filter" Ideal filter with no frequencies above cutoff (infinite extent!)

$$sinc(x) = \frac{sin(\pi x))}{\pi x}$$



0.2 -0.2

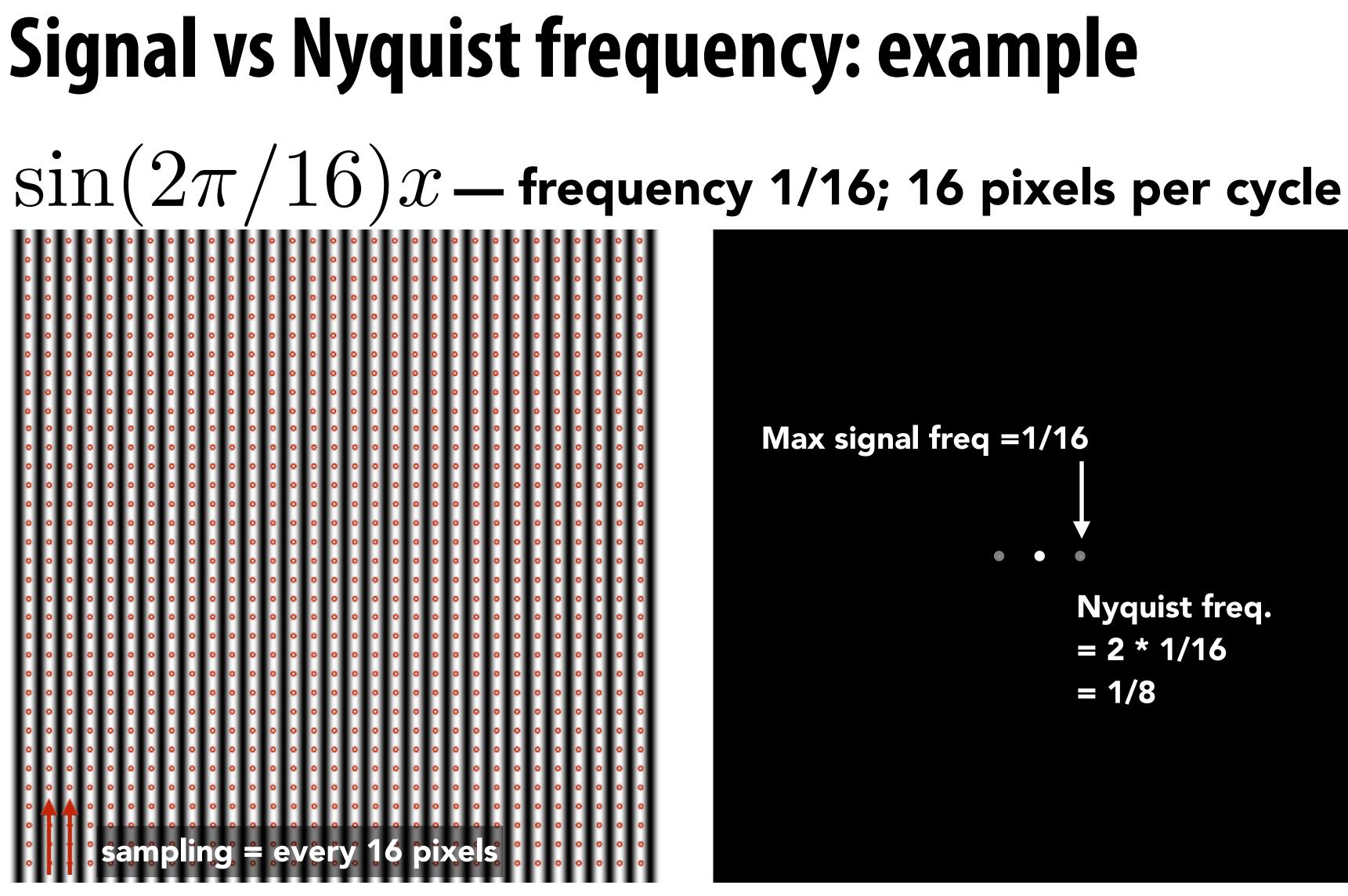
Stantora (S248, Winter 2021



Spatial domain



Frequency domain



Aliasing! (due to undersampling)

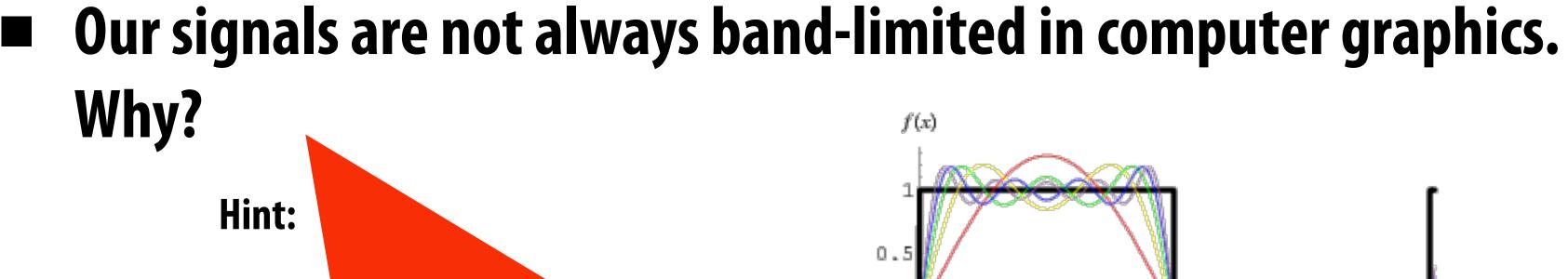
Reminder: Nyquist theorem

Theorem: We get no aliasing from frequencies in the signal that are less than the Nyquist frequency (which is defined as half the sampling frequency)

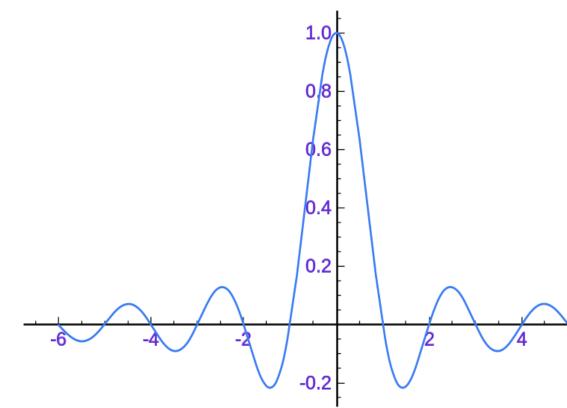
Consequence: sampling at twice the highest frequency in the signal will eliminate aliasing

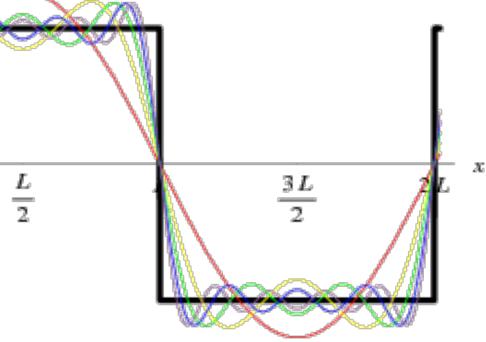
Challenges of sampling-based approaches in graphics

-0.5



Also, infinite extent of "ideal" reconstruction filter (sinc) is impractical for efficient implementations. Why?







Acknowledgements

Thanks to Ren Ng, Pat Hanrahan, Keenan Crane for slide materials