

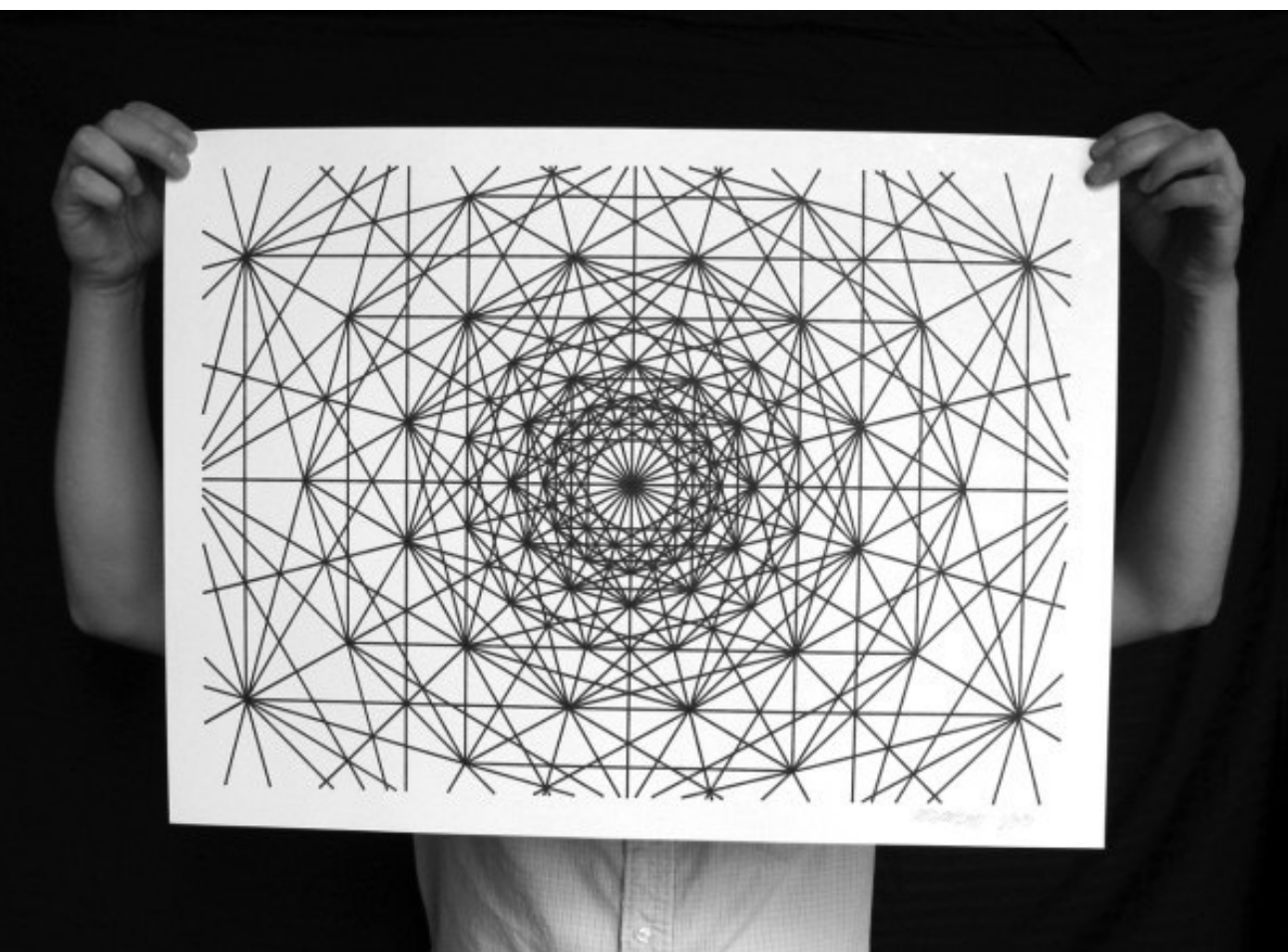
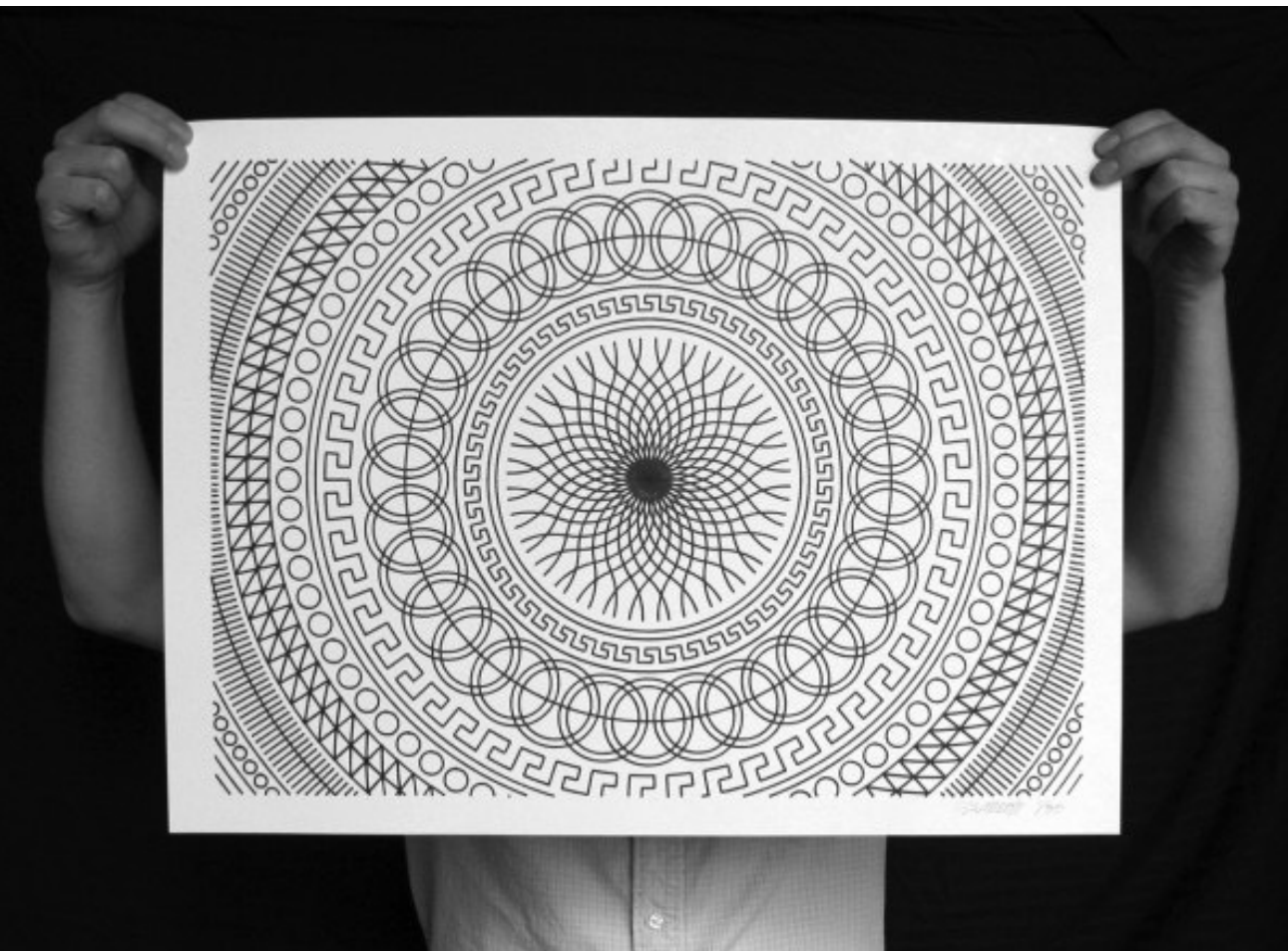
**Lecture 2:**

# **Drawing a Triangle** **(+ the basics of sampling/anti-aliasing)**

---

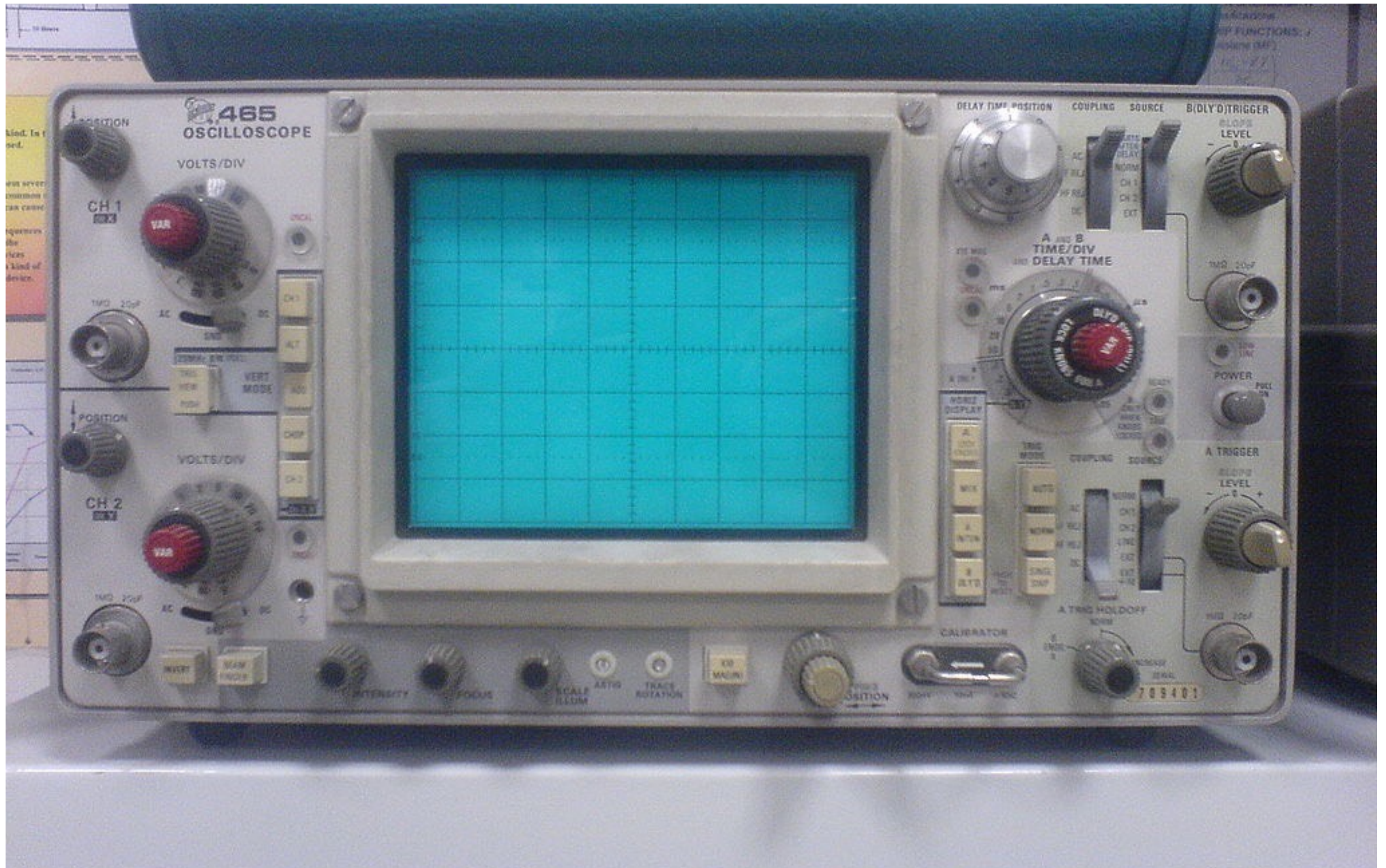
**Interactive Computer Graphics**  
**Stanford CS248, Spring 2018**

# CNC sharpie drawing machine ;-)

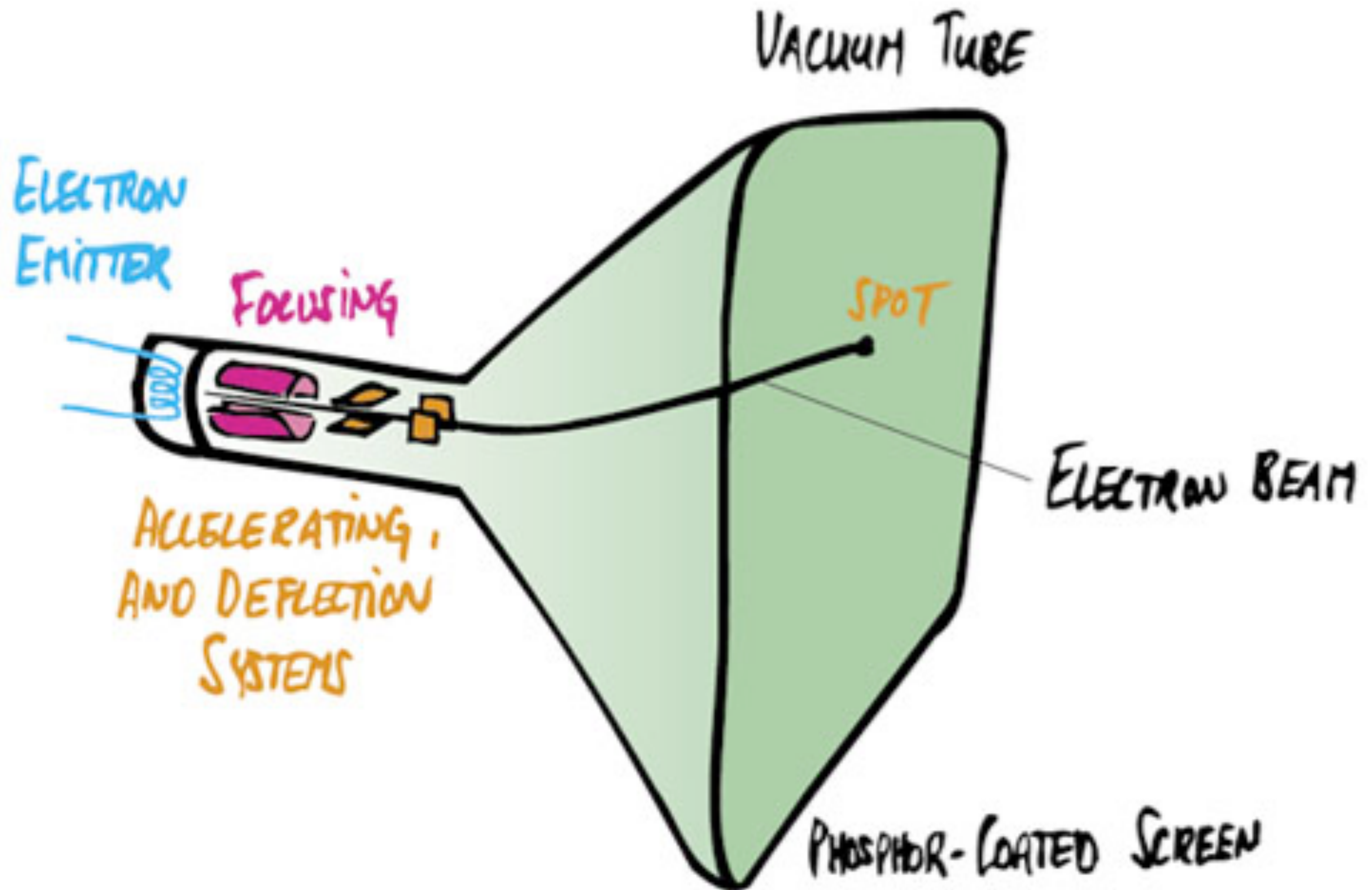


<http://44rn.com/projects/numerically-controlled-poster-series-with-matt-w-moore/>

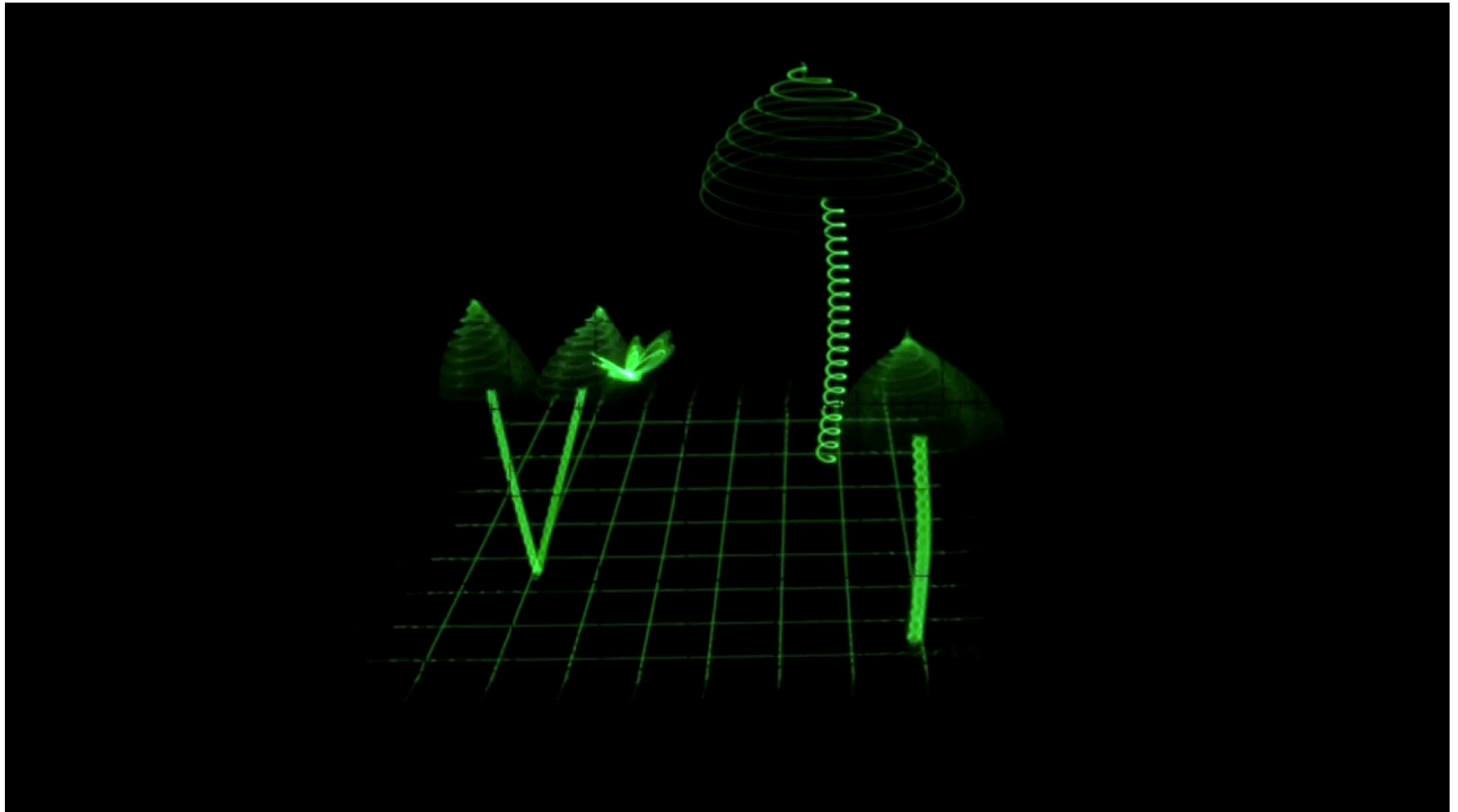
# Oscilloscope



# Cathode ray tube



# Oscilloscope art



<https://www.youtube.com/watch?v=rtR63-ecUNo>

# Frame buffer: memory for a raster display



**image =**  
**“2D array of colors”**

# Flat panel displays



Low-Res LCD Display

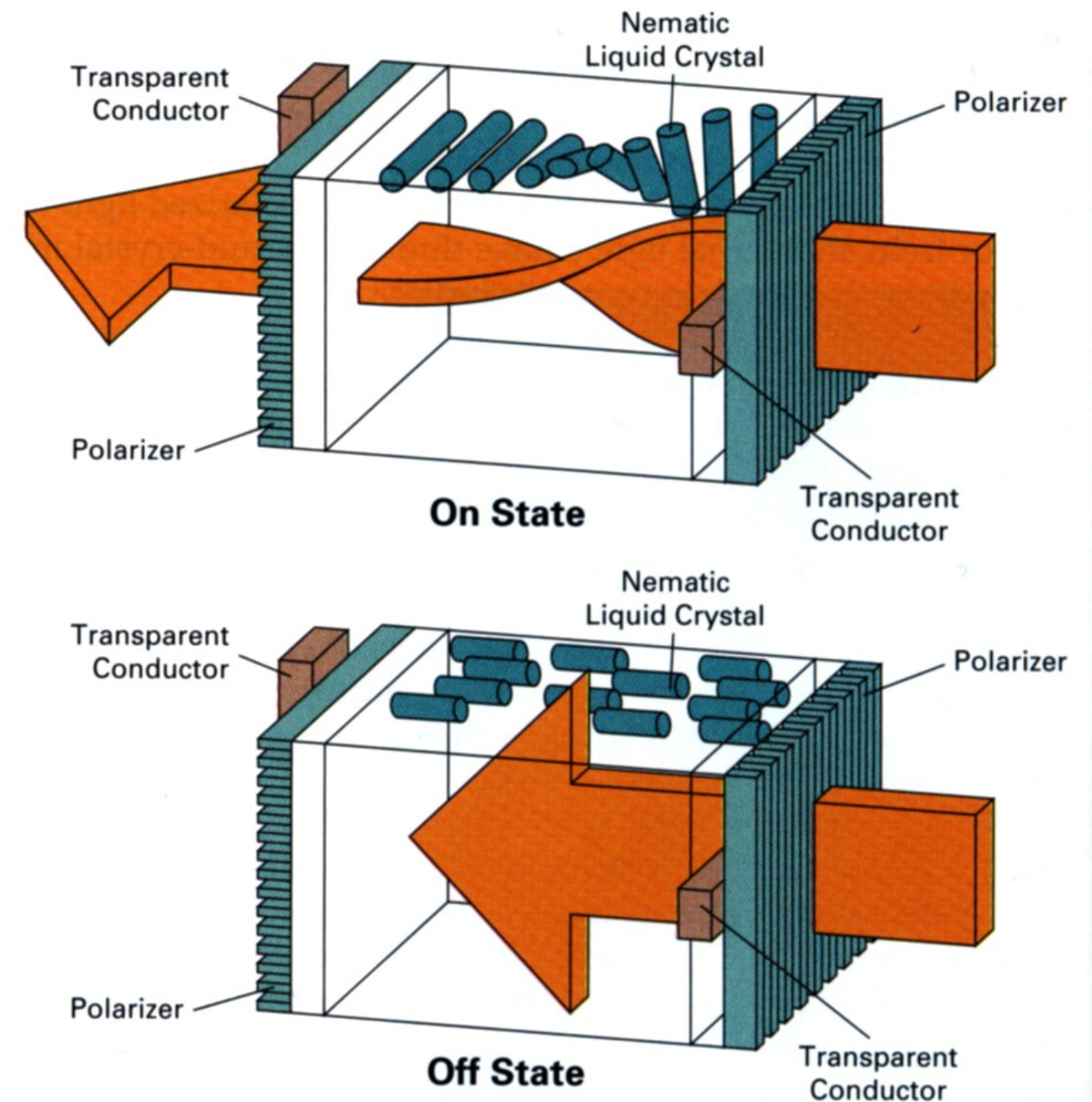


High resolution color LCD, OLED, ...

B.Woods, Android Pit

# LCD (liquid crystal display) pixel

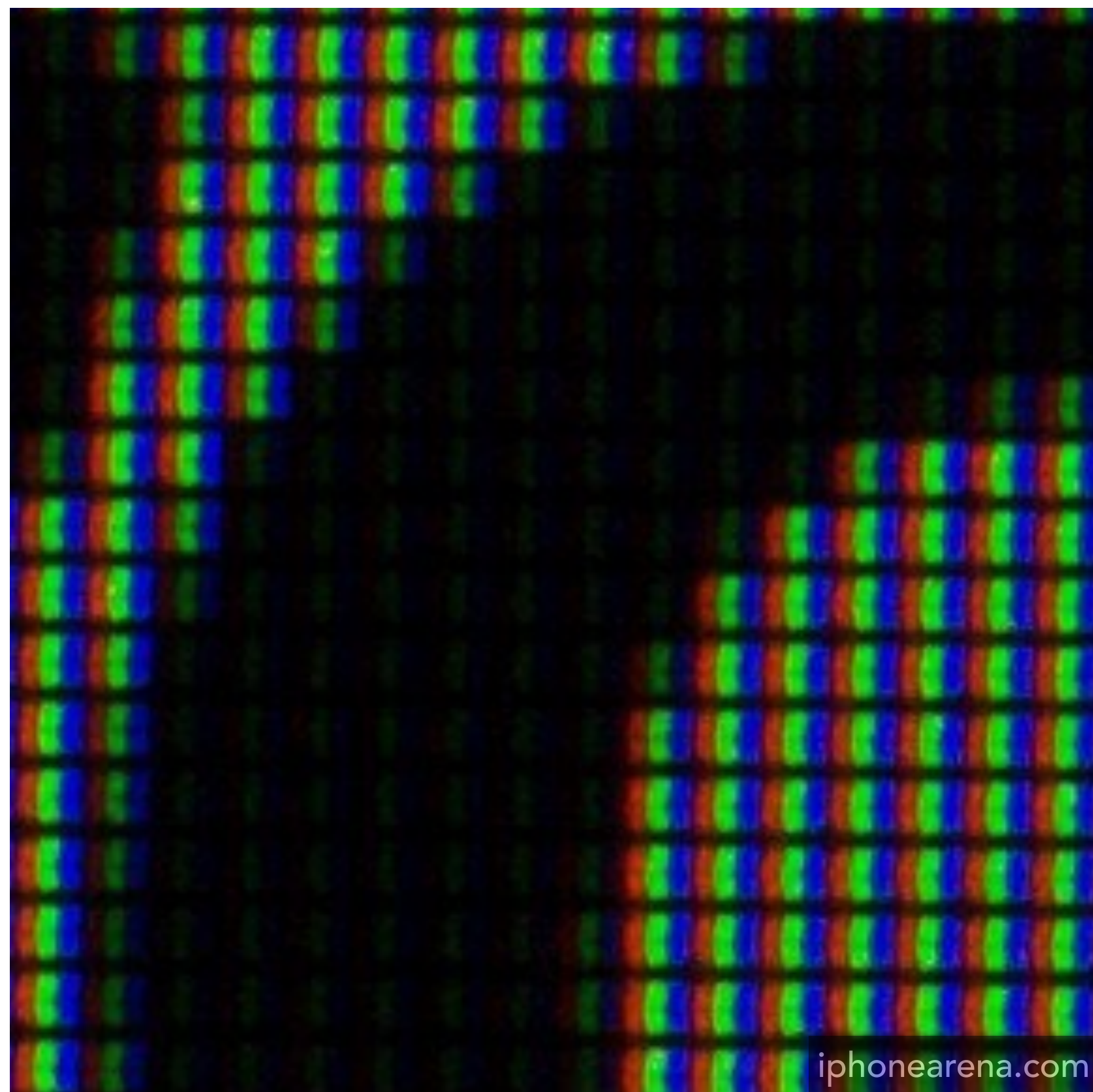
- Principle: block or transmit light by twisting polarization
- Illumination from backlight (e.g. fluorescent or LED)
- Intermediate intensity levels by partial twist



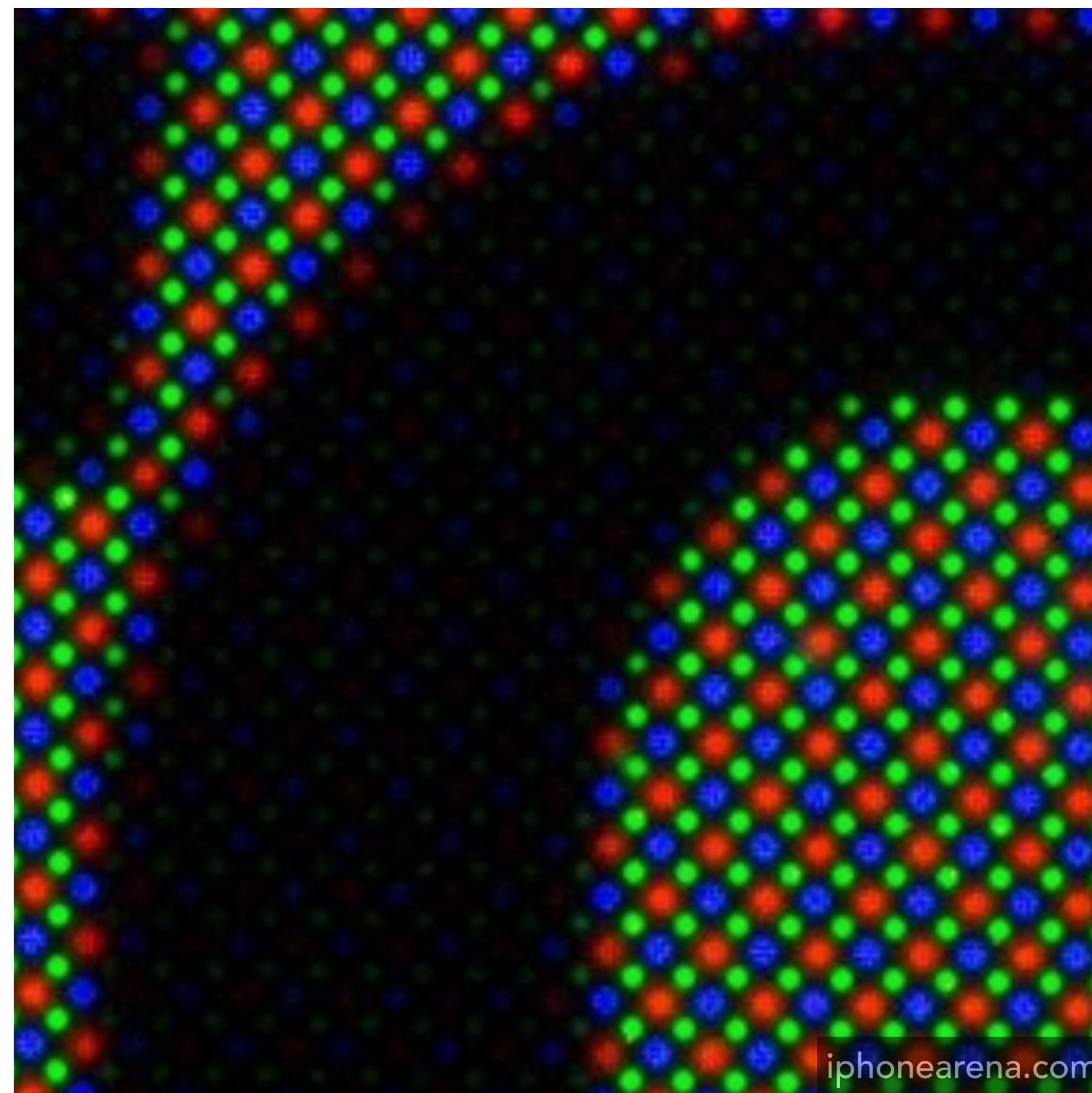
[H&B fig. 2-16]



# LCD screen pixels (closeup)



**iPhone 6S**



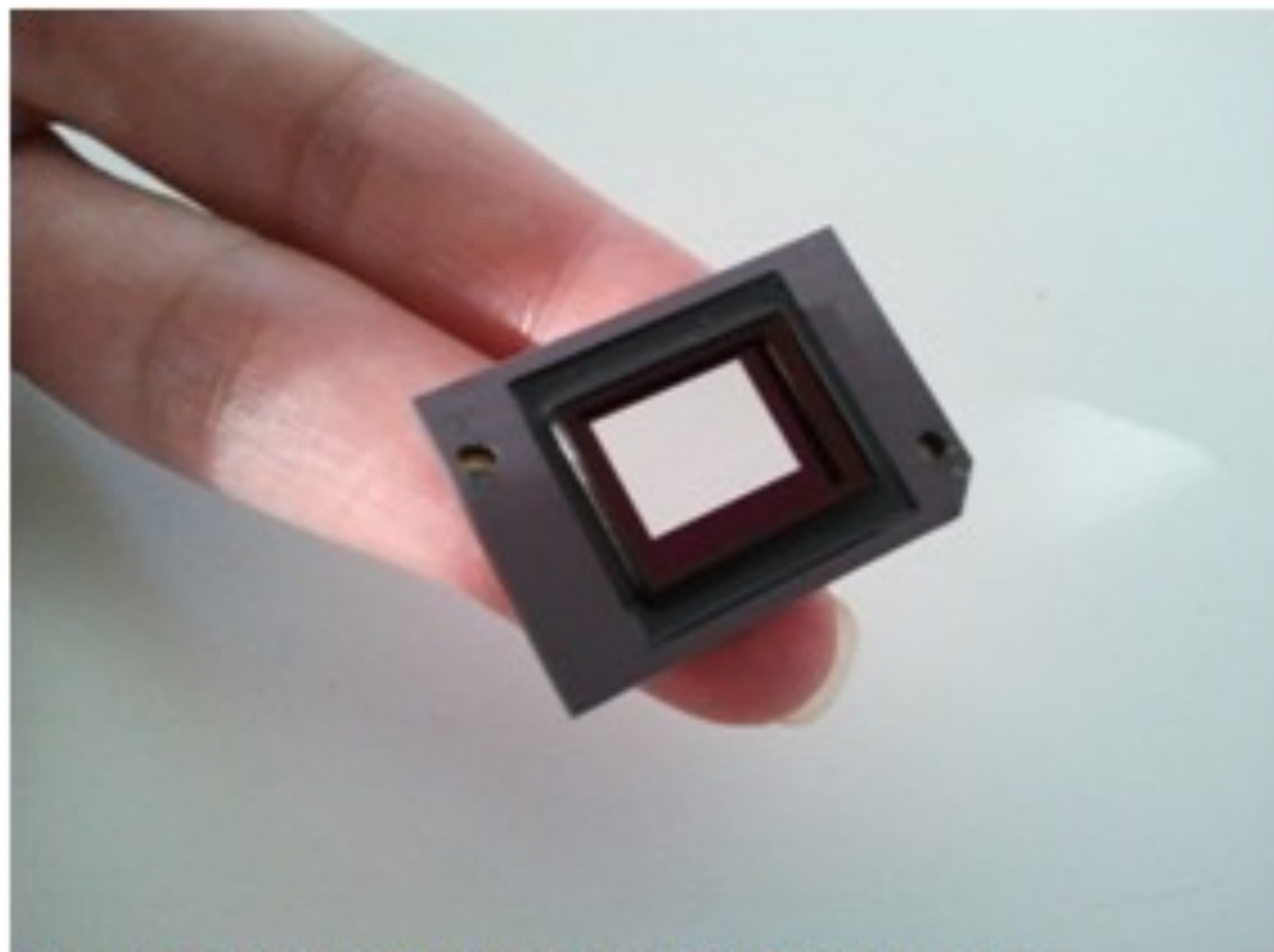
**Galaxy S5**

# LED array display

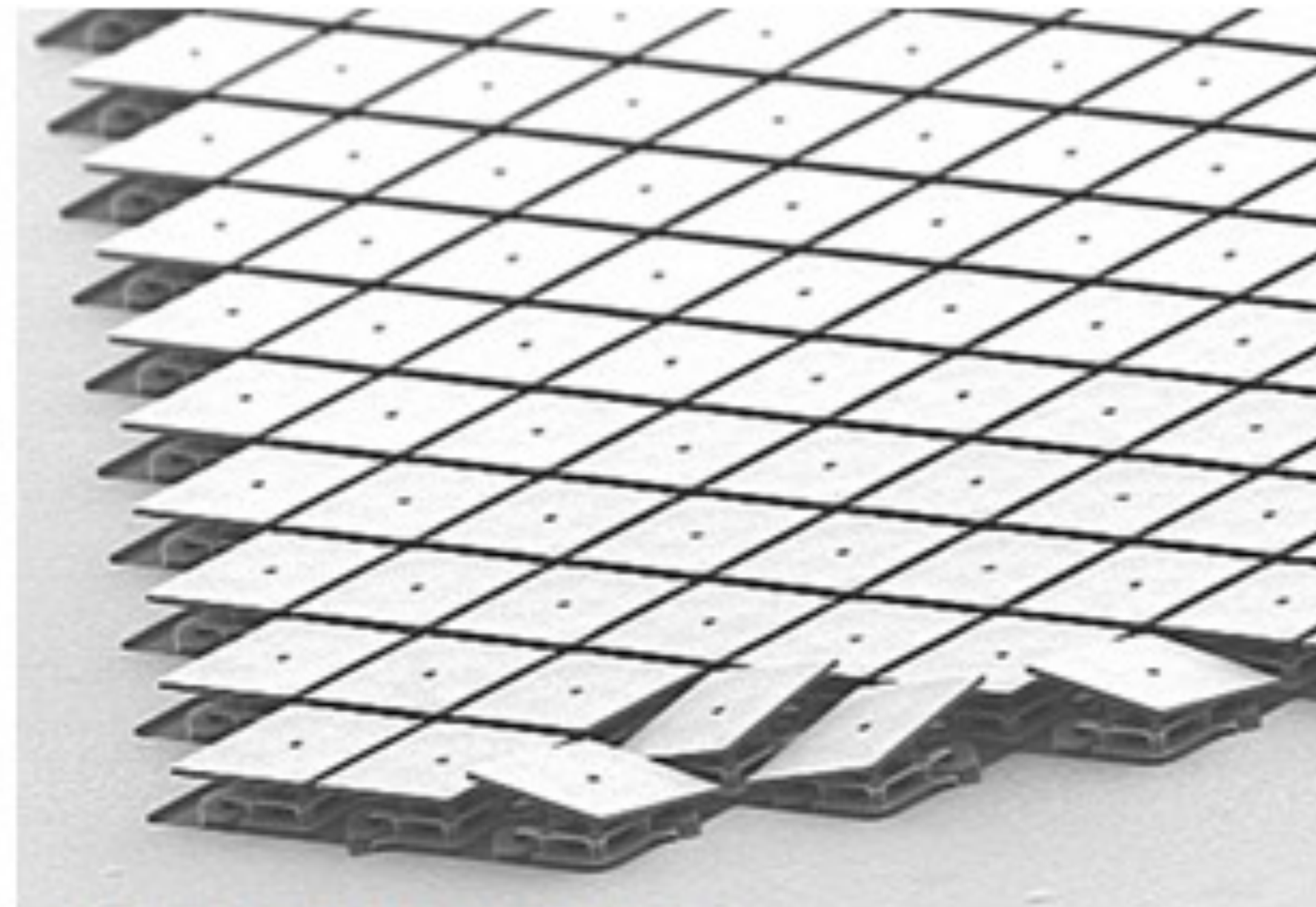


**Light emitting diode array**

# DMD projection display



DIGITAL MICRO MIRROR DEVICE (**DMD**)  
(**SLM** - Spatial Light Modulator)



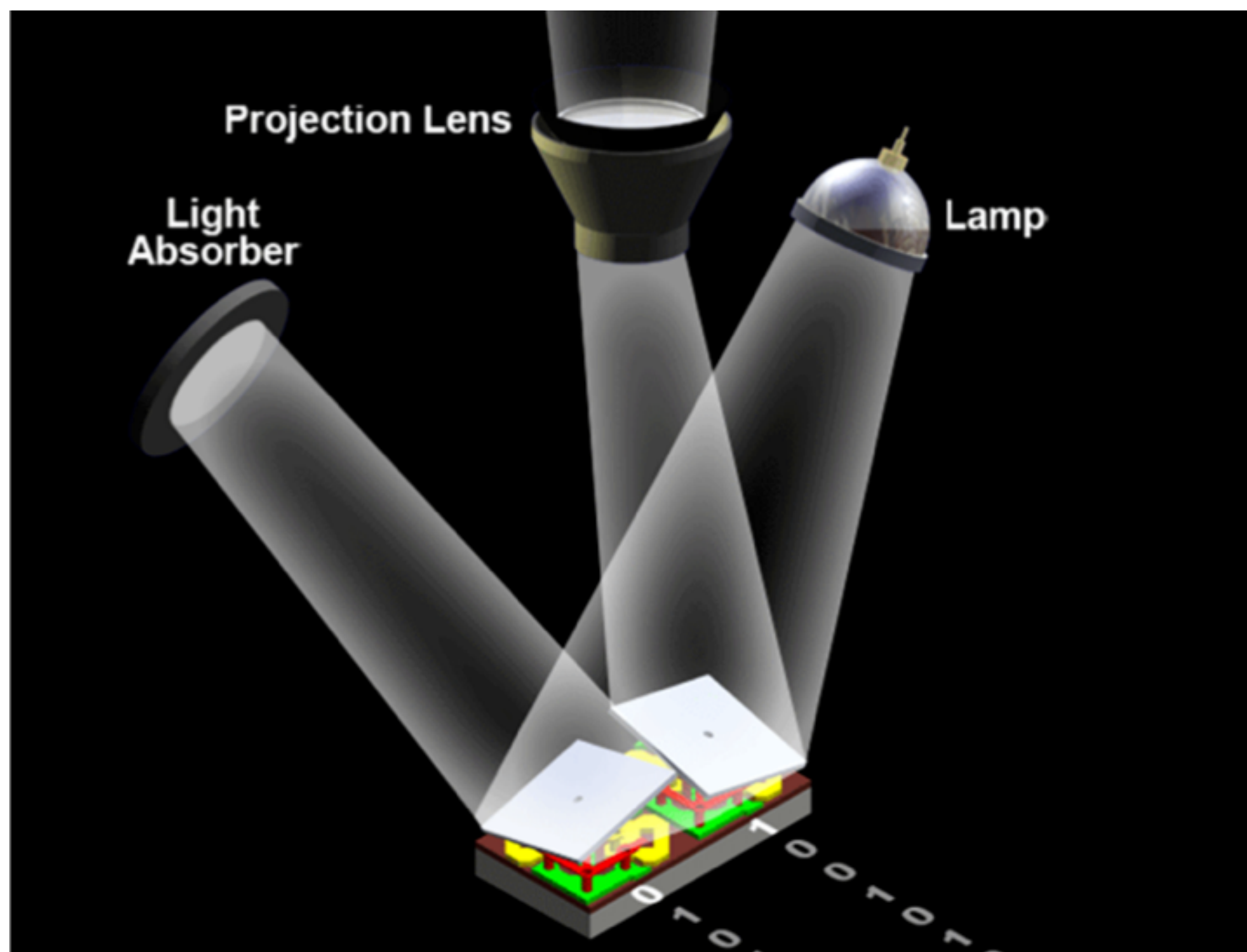
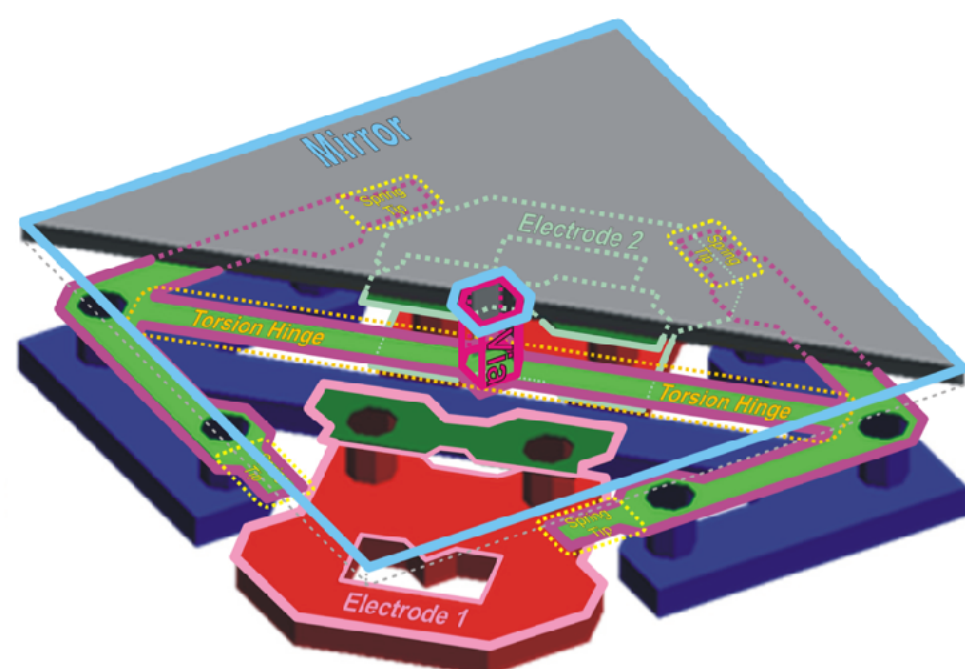
MICRO MIRRORS CLOSE UP

[Y.K. Rabinowitz; EKB Technologies

**Array of micro-mirror pixels**

**DMD = Digital micro-mirror device**

# DMD projection display



[Texas Instruments]

**Array of micro-mirror pixels**

**DMD = Digital micro-mirror device**

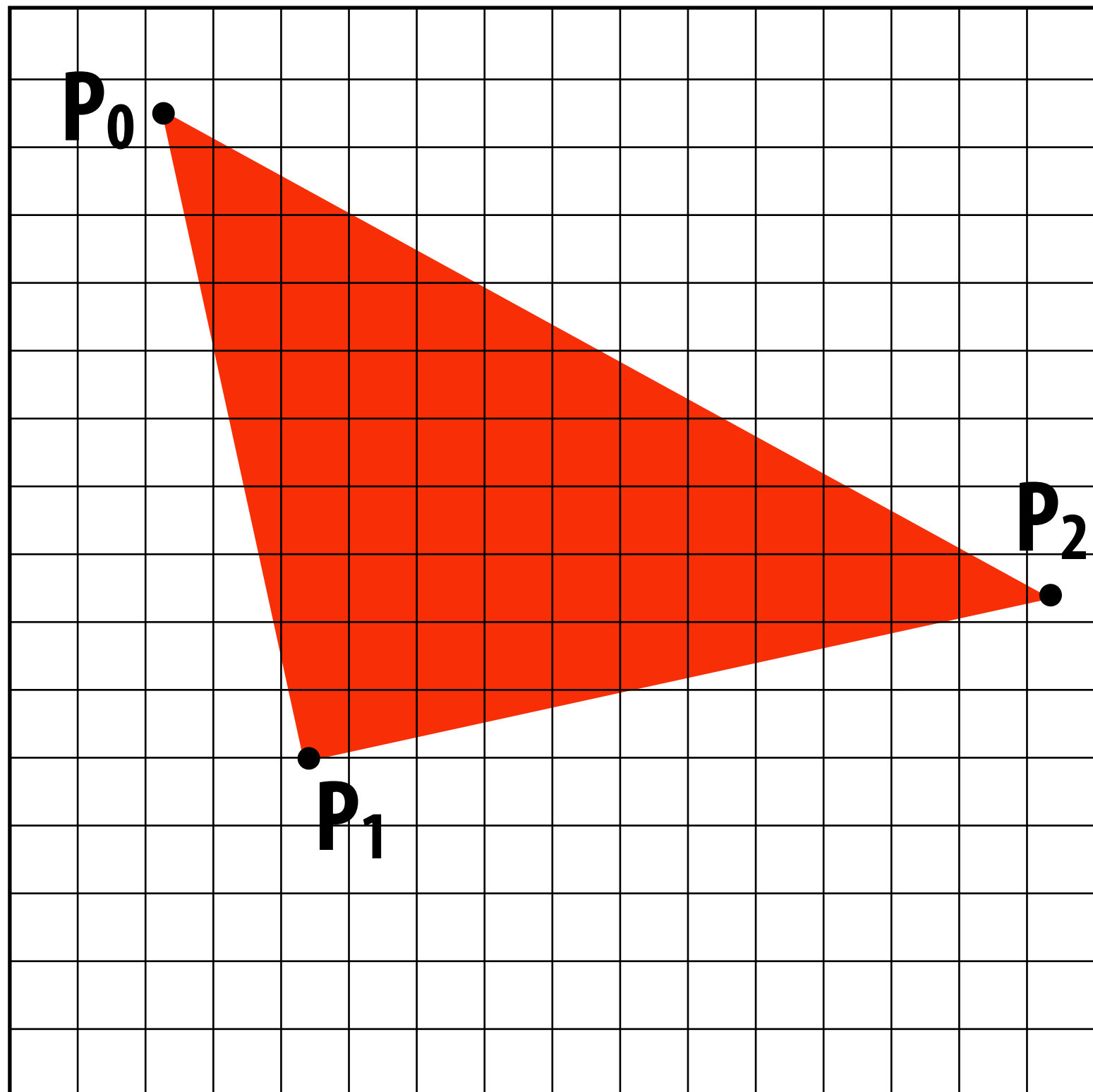
# **Drawing a triangle to a frame buffer (triangle “rasterization”)**

# Today: drawing a triangle to a frame buffer

Determining what pixels the triangle overlaps?

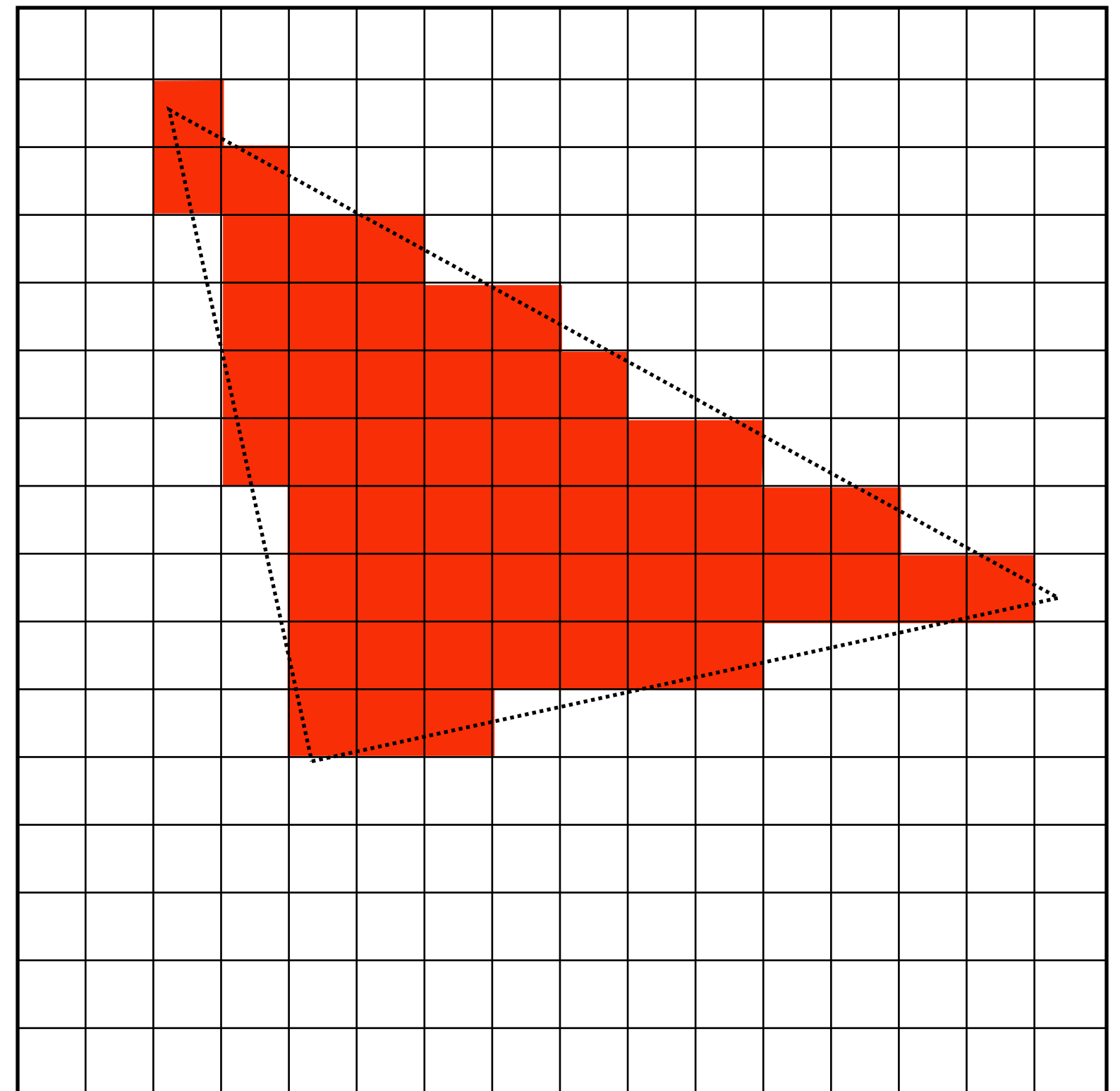
**Input:**

projected position of triangle vertices:  $P_0, P_1, P_2$



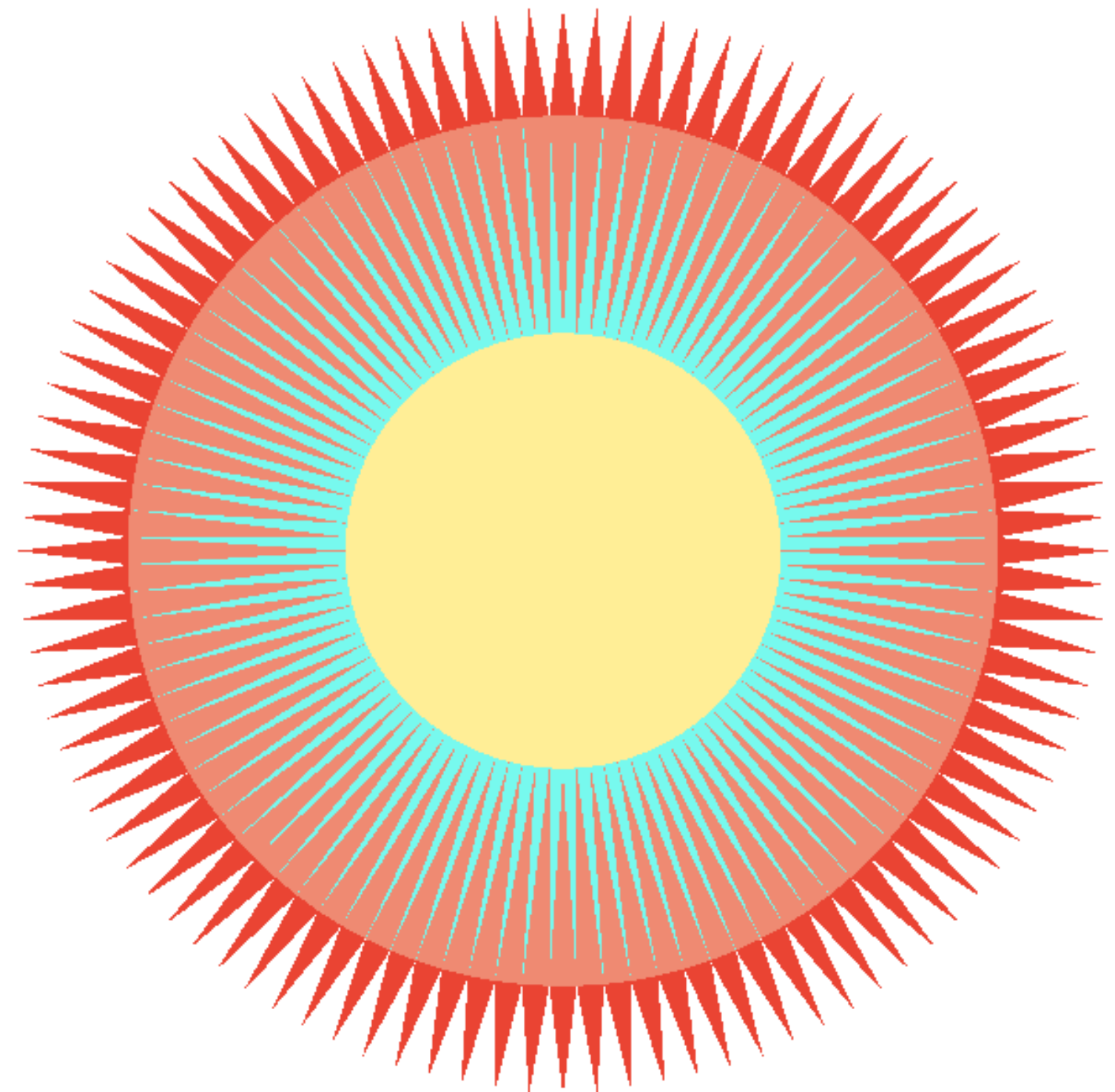
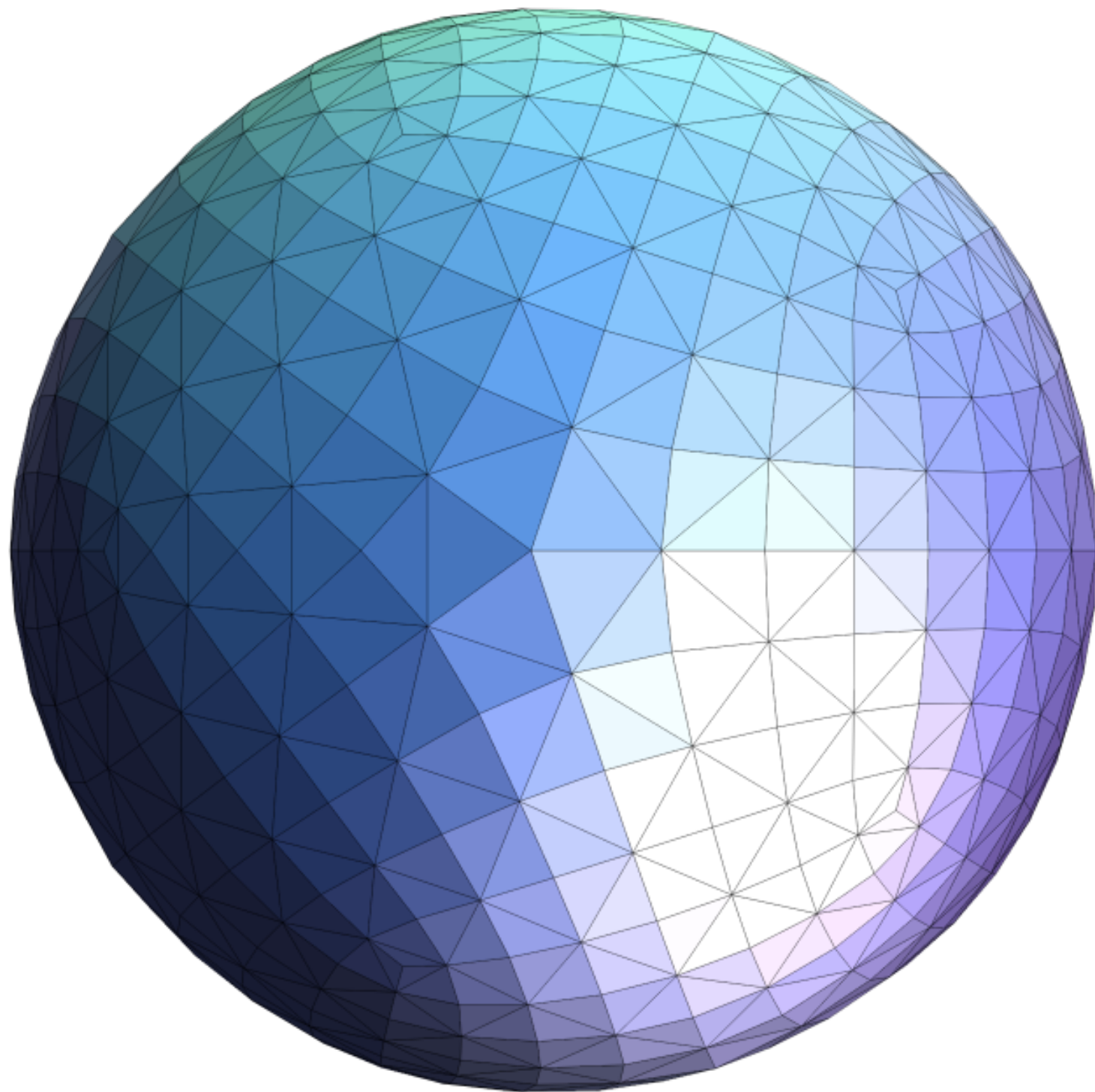
**Output:**

set of pixels "covered" by the triangle



# Why triangles?

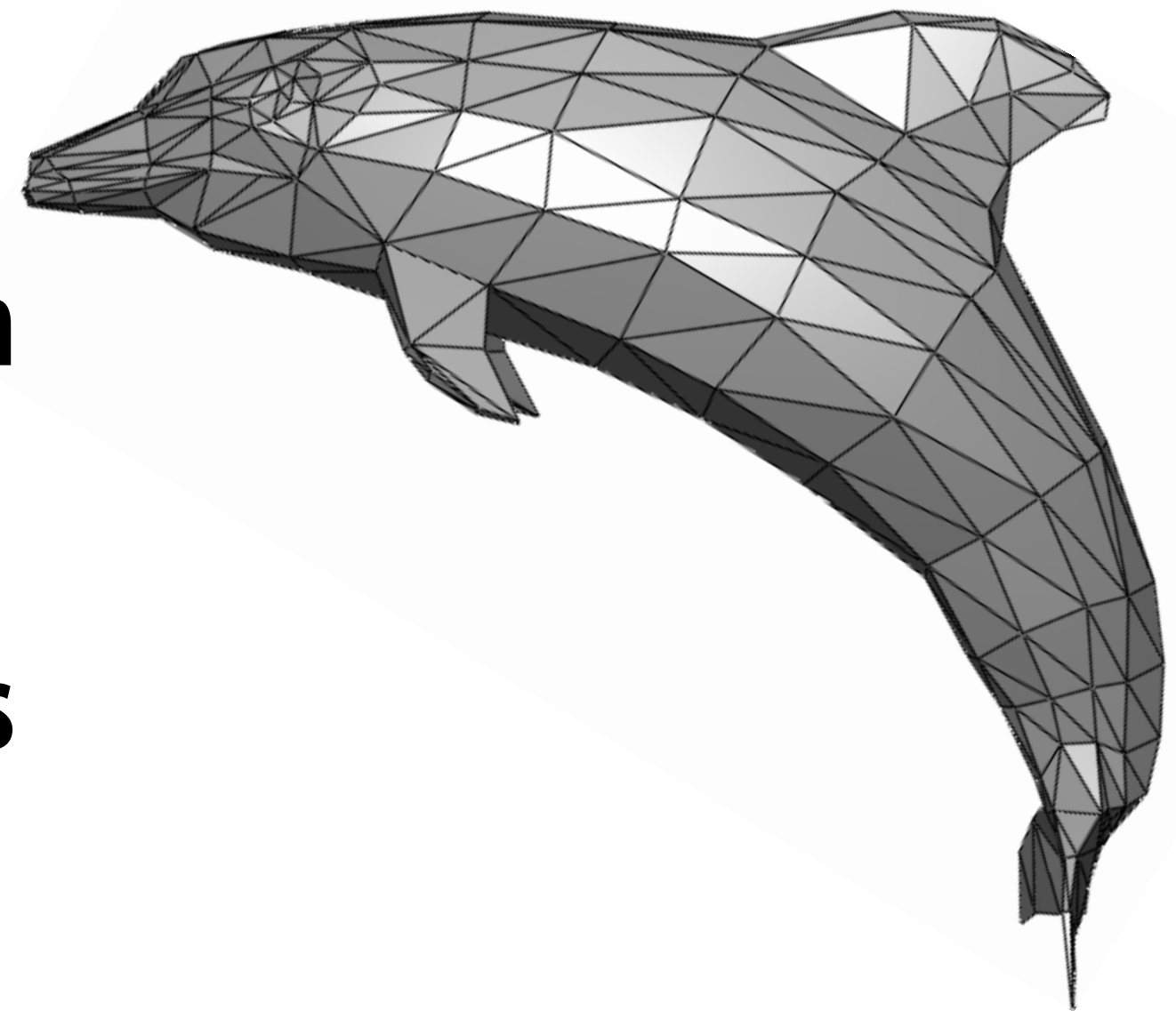
**Triangles are a basic block for creating more complex shapes and surfaces**



# Triangles - fundamental primitive

## ■ Why triangles?

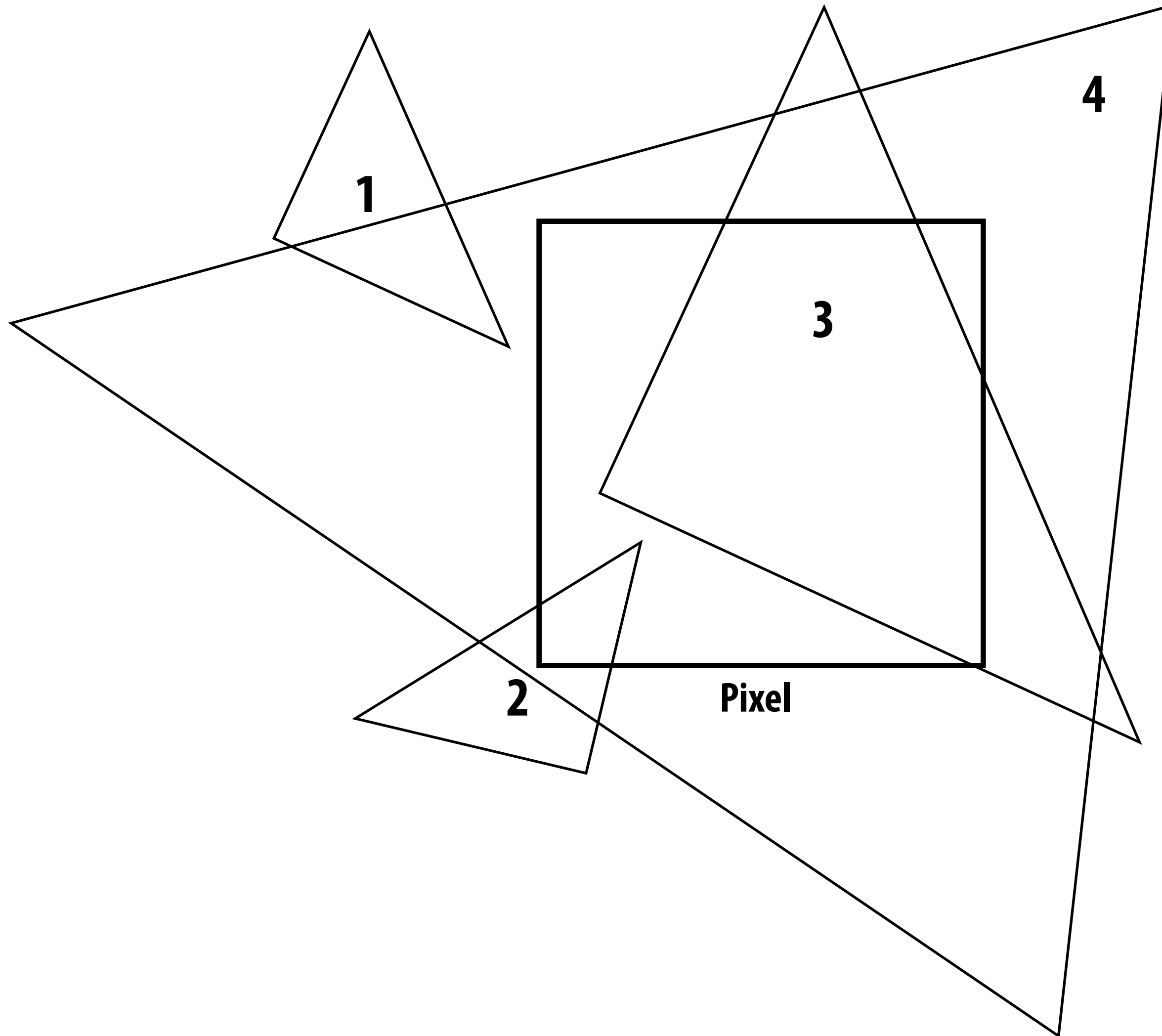
- **Most basic polygon**
  - **Break up other polygons**
  - **Optimize one implementation**
- **Triangles have unique properties**
  - **Guaranteed to be planar**
  - **Well-defined interior**
  - **Well-defined method for interpolating values at vertices over triangle (barycentric interpolation)**



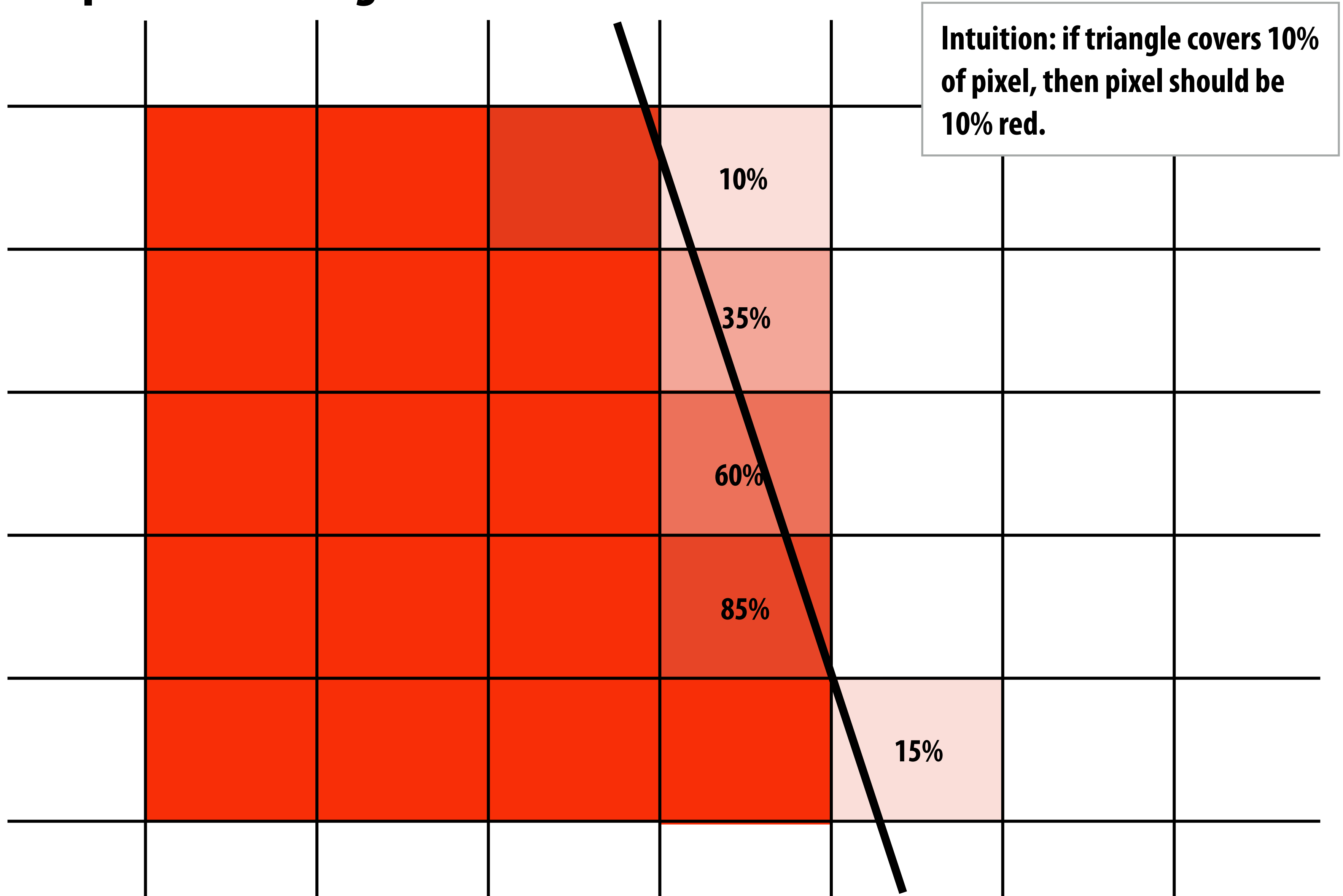


# What does it mean for a pixel to be covered by a triangle?

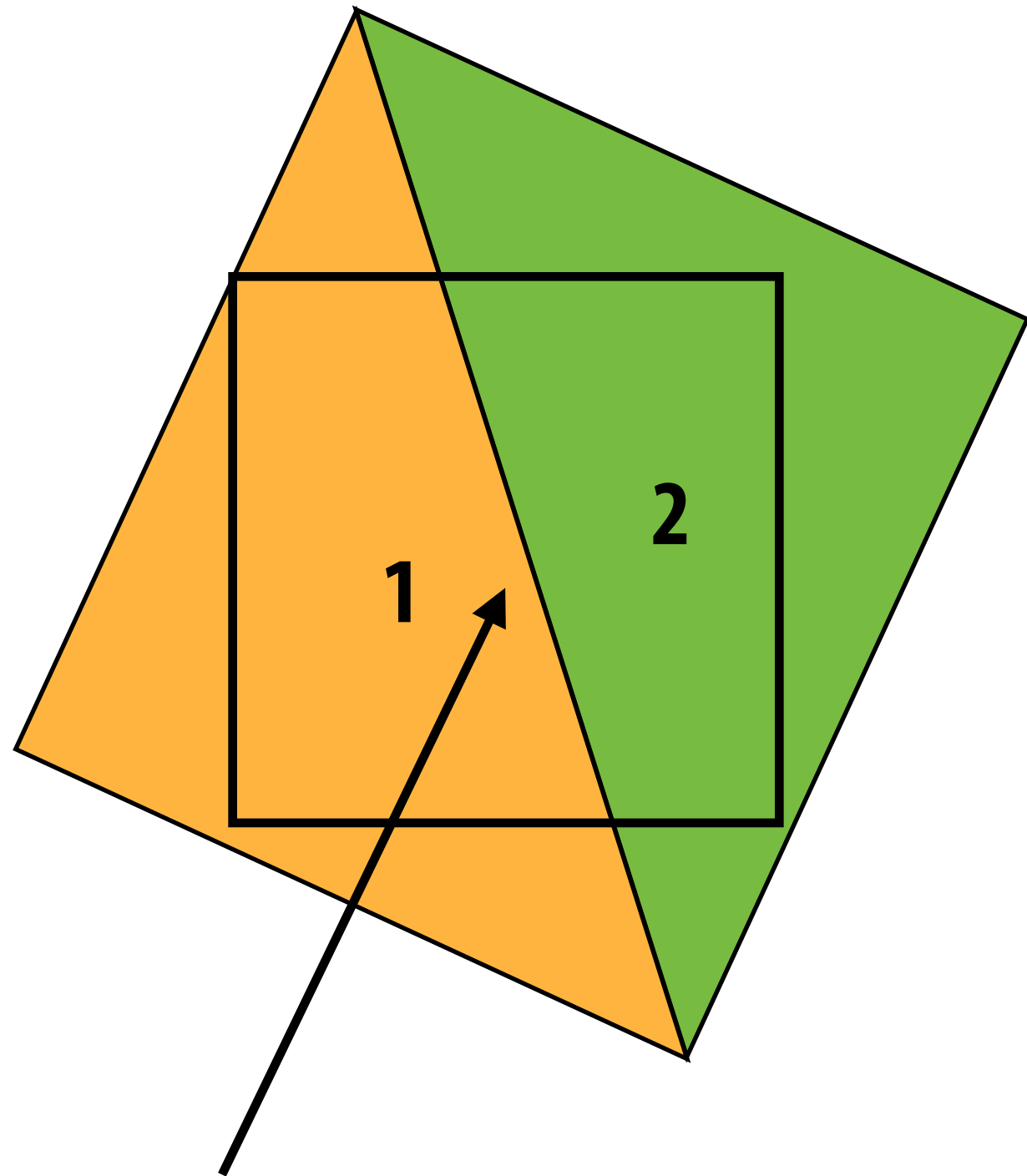
Question: which triangles "cover" this pixel?



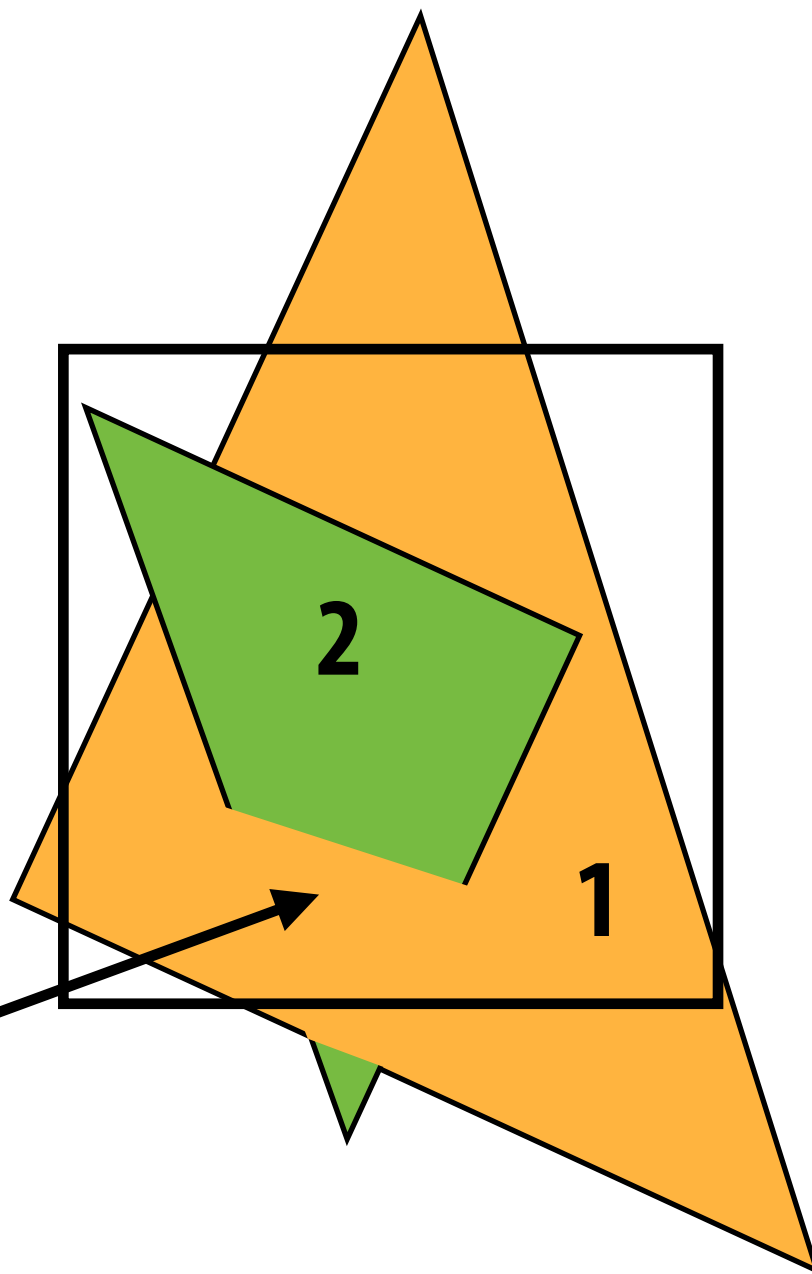
**One option: compute fraction of pixel area covered by triangle, then color pixel according to this fraction.**



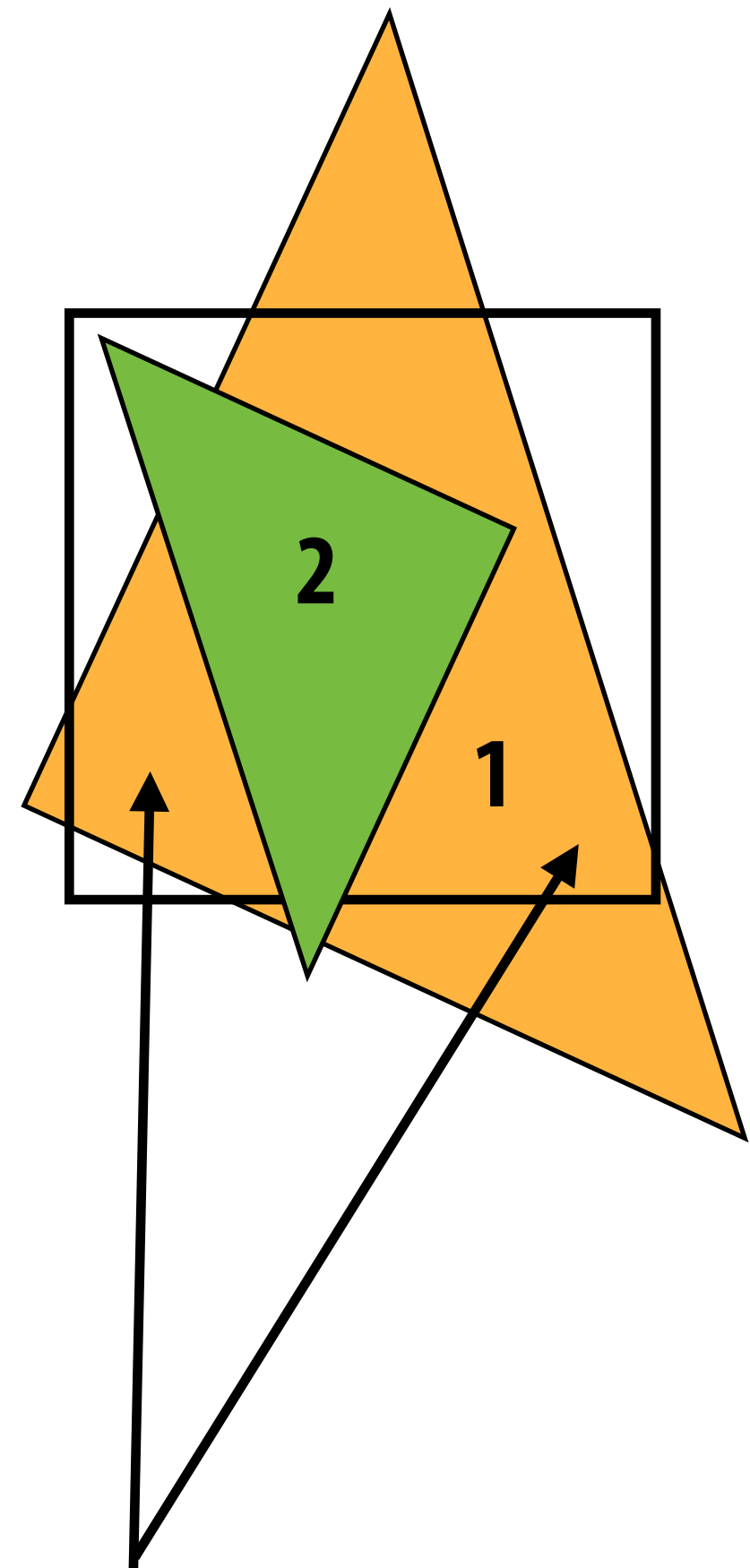
# Analytical coverage schemes get tricky when considering occlusion of one triangle by another



Pixel covered by triangle 1, other half covered by triangle 2



Interpenetration of triangles: even trickier

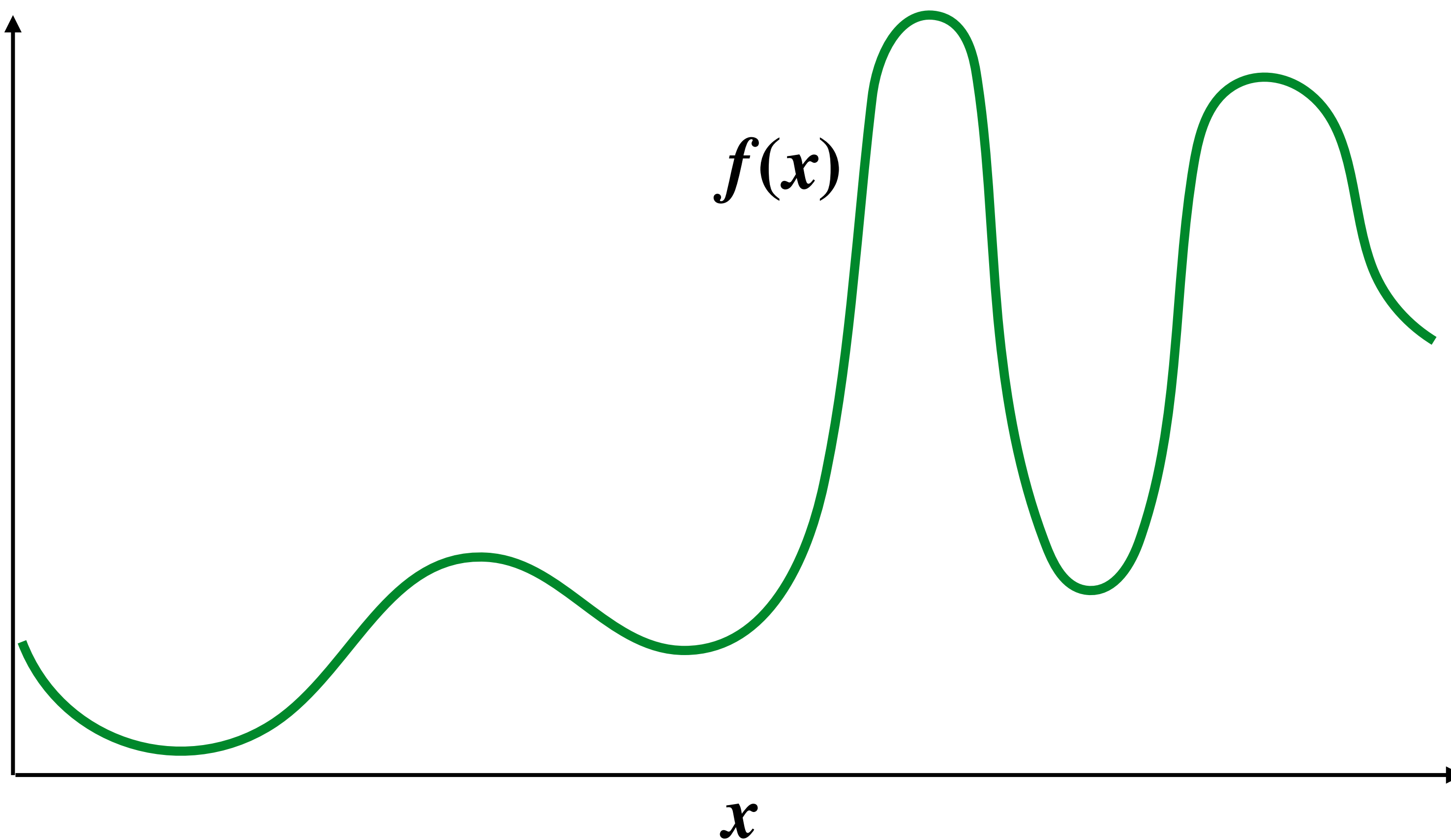


Two regions of triangle 1 contribute to pixel. One of these regions is not even convex.

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

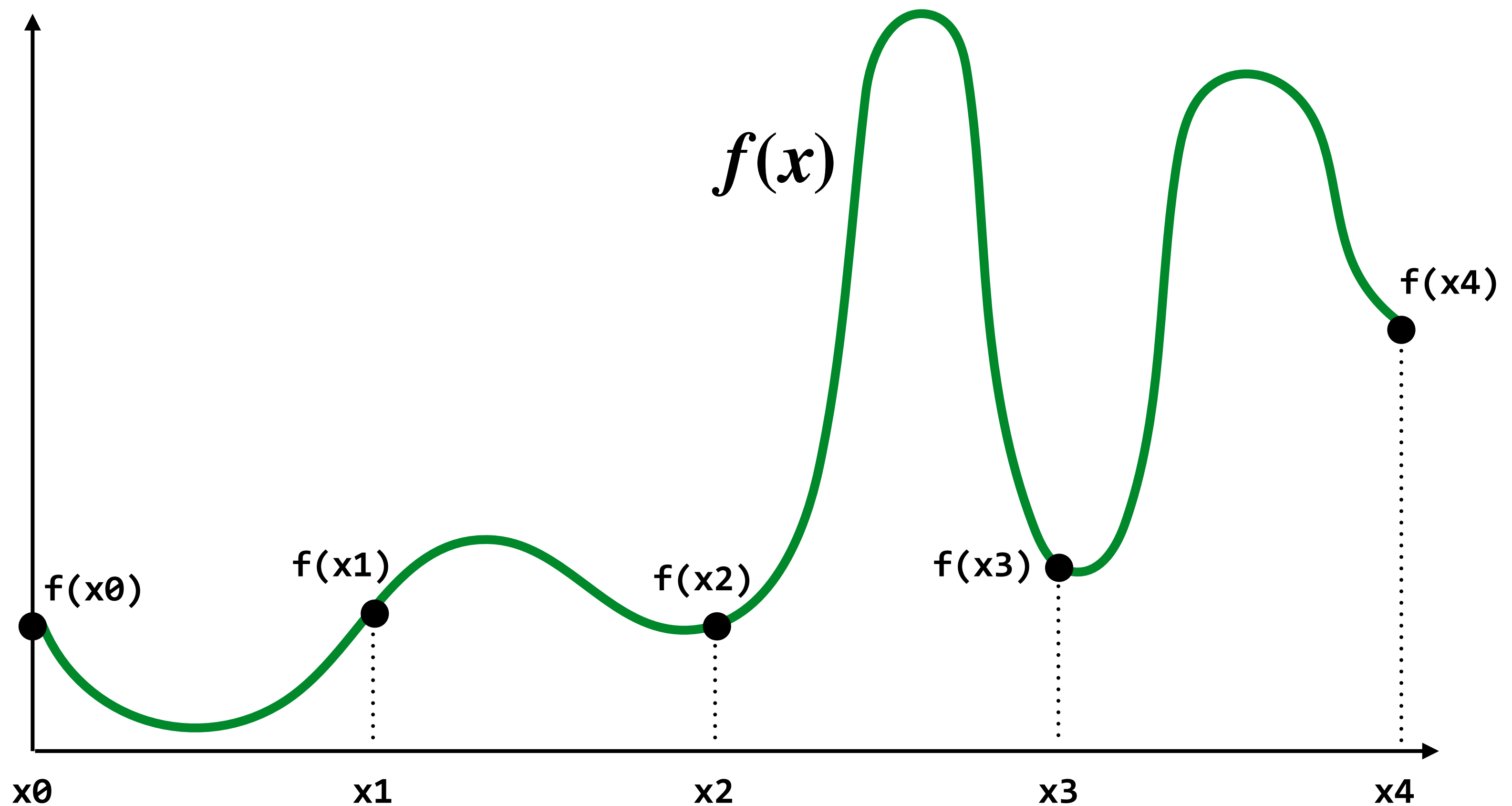
**(let's consider sampling in 1D first)**

# Consider a 1D signal: $f(x)$



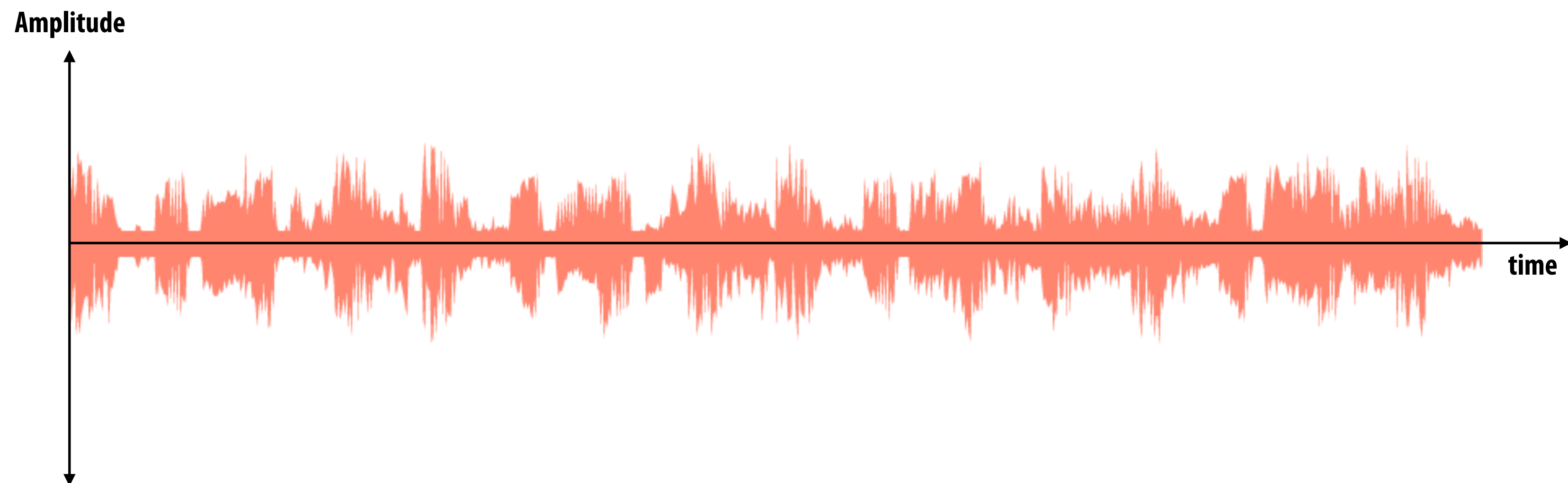
# Sampling: taking measurements a signal

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



# Audio file: stores samples of a 1D signal

Most consumer audio is sampled at 44.1 KHz



# Sampling a function

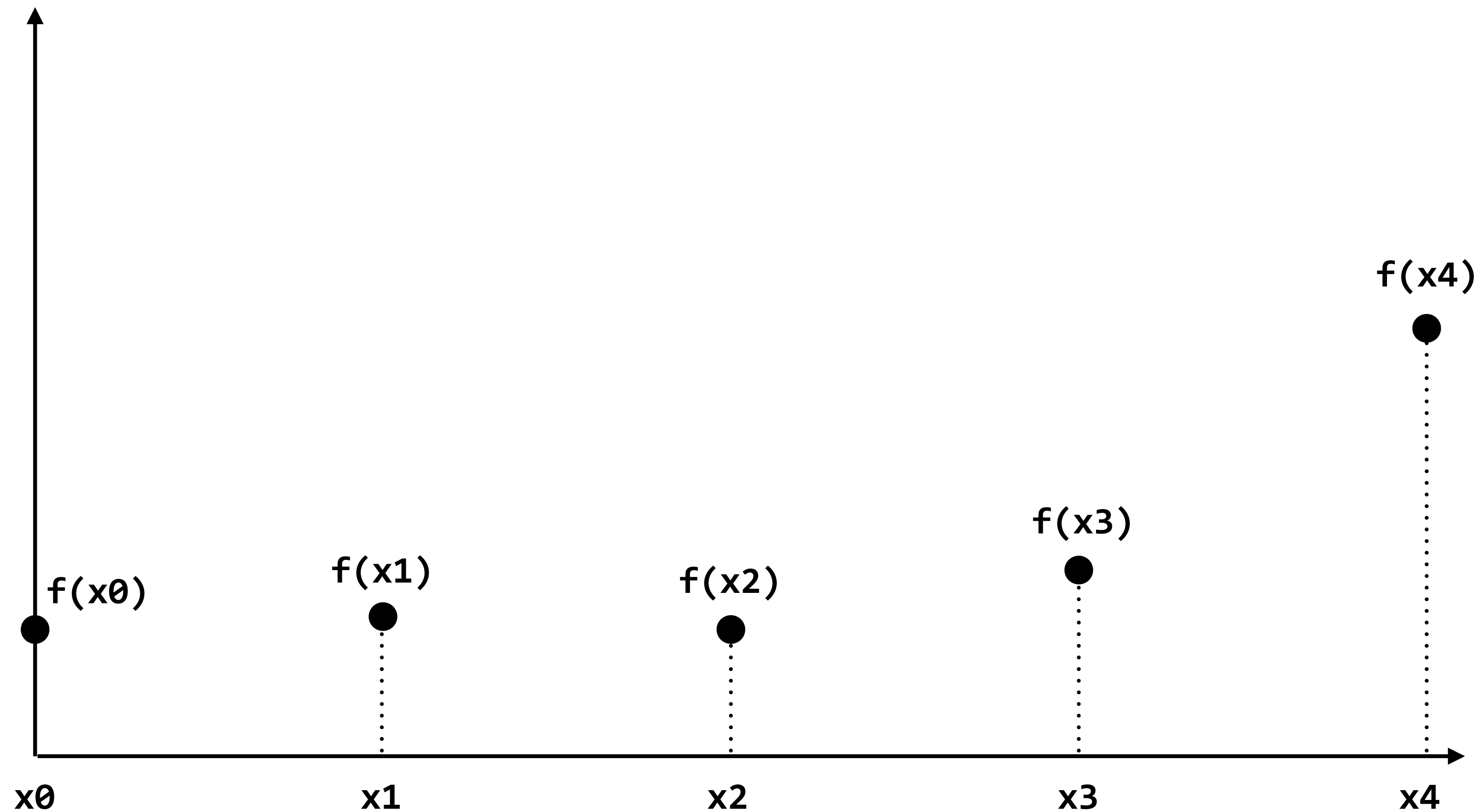
- **Evaluating a function at a point is sampling**
- **We can discretize a function by periodic sampling**

```
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 ...**



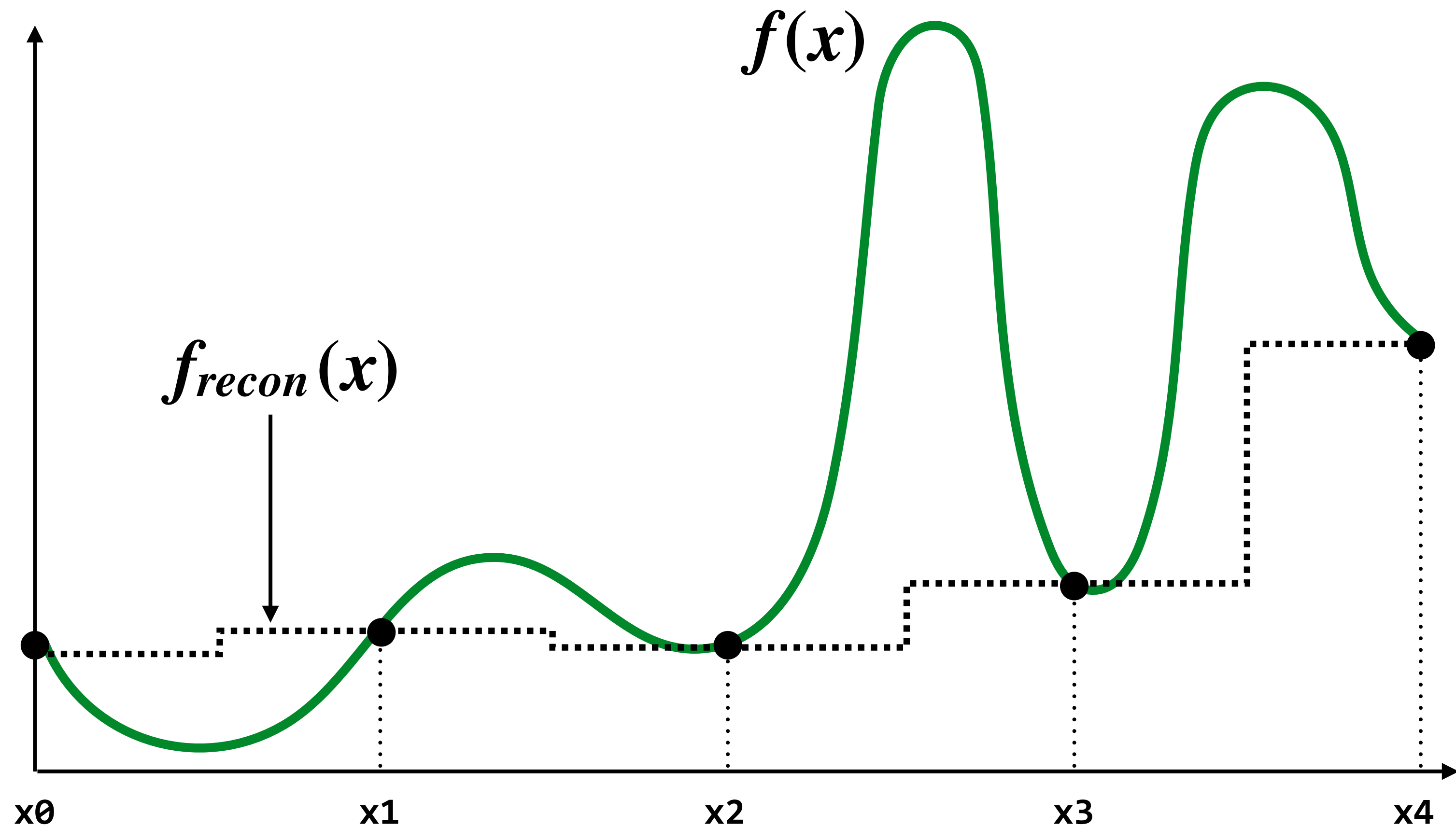
**Reconstruction: given a set of samples, how might we attempt to reconstruct the original signal  $f(x)$ ?**



# Piecewise constant approximation

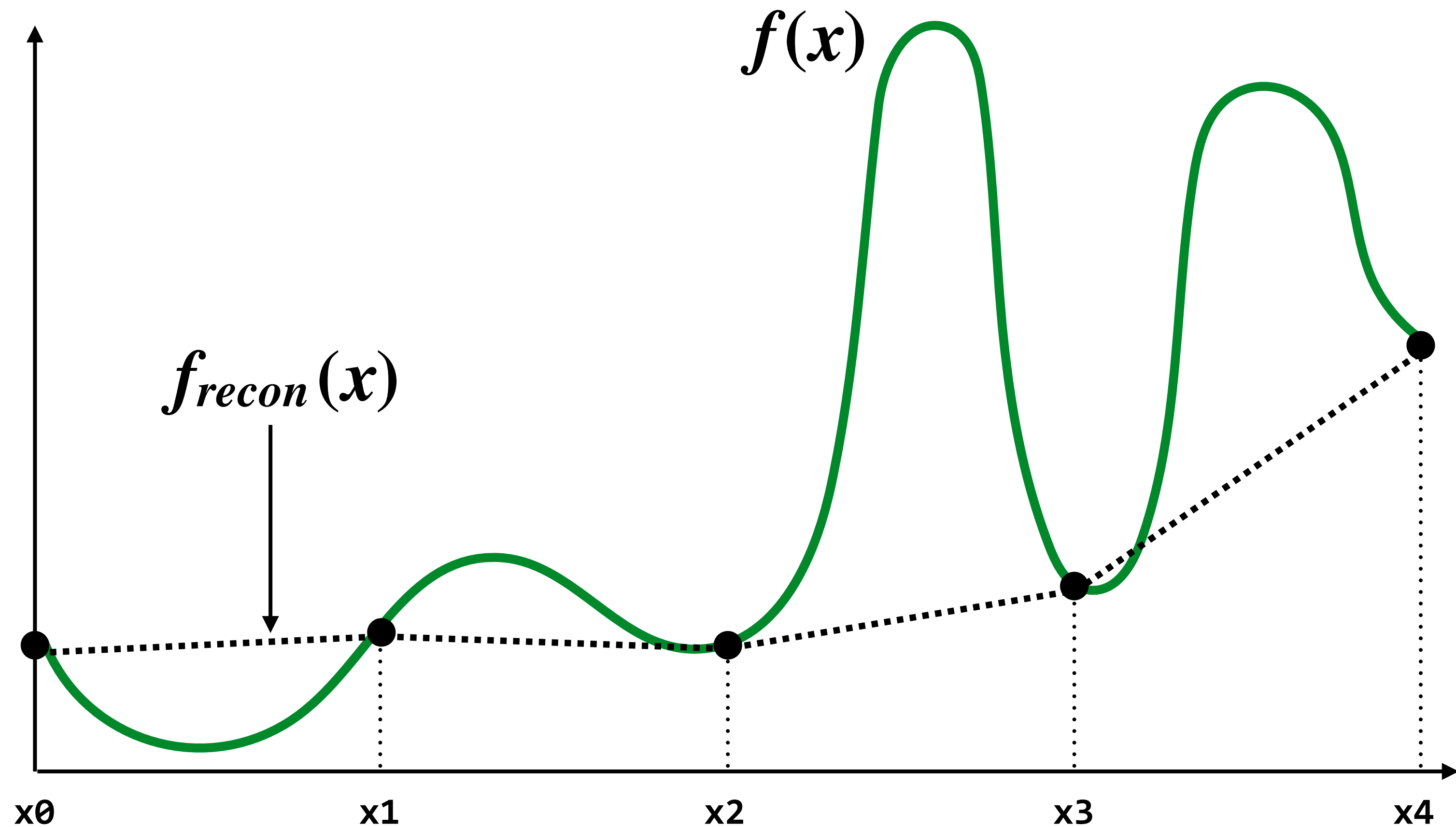
$f_{recon}(x)$  = value of sample closest to  $x$

$f_{recon}(x)$  approximates  $f(x)$

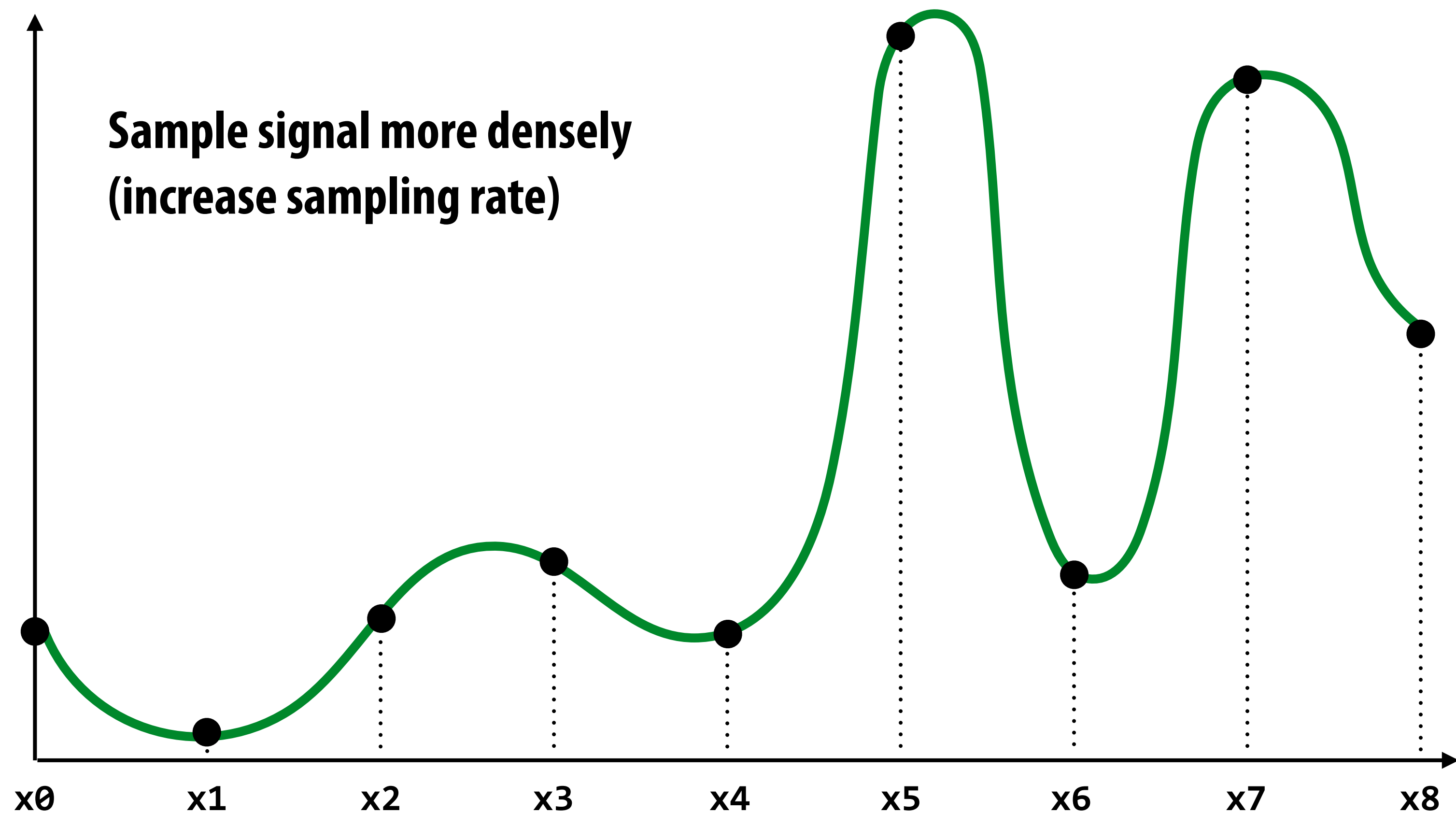


# Piecewise linear approximation

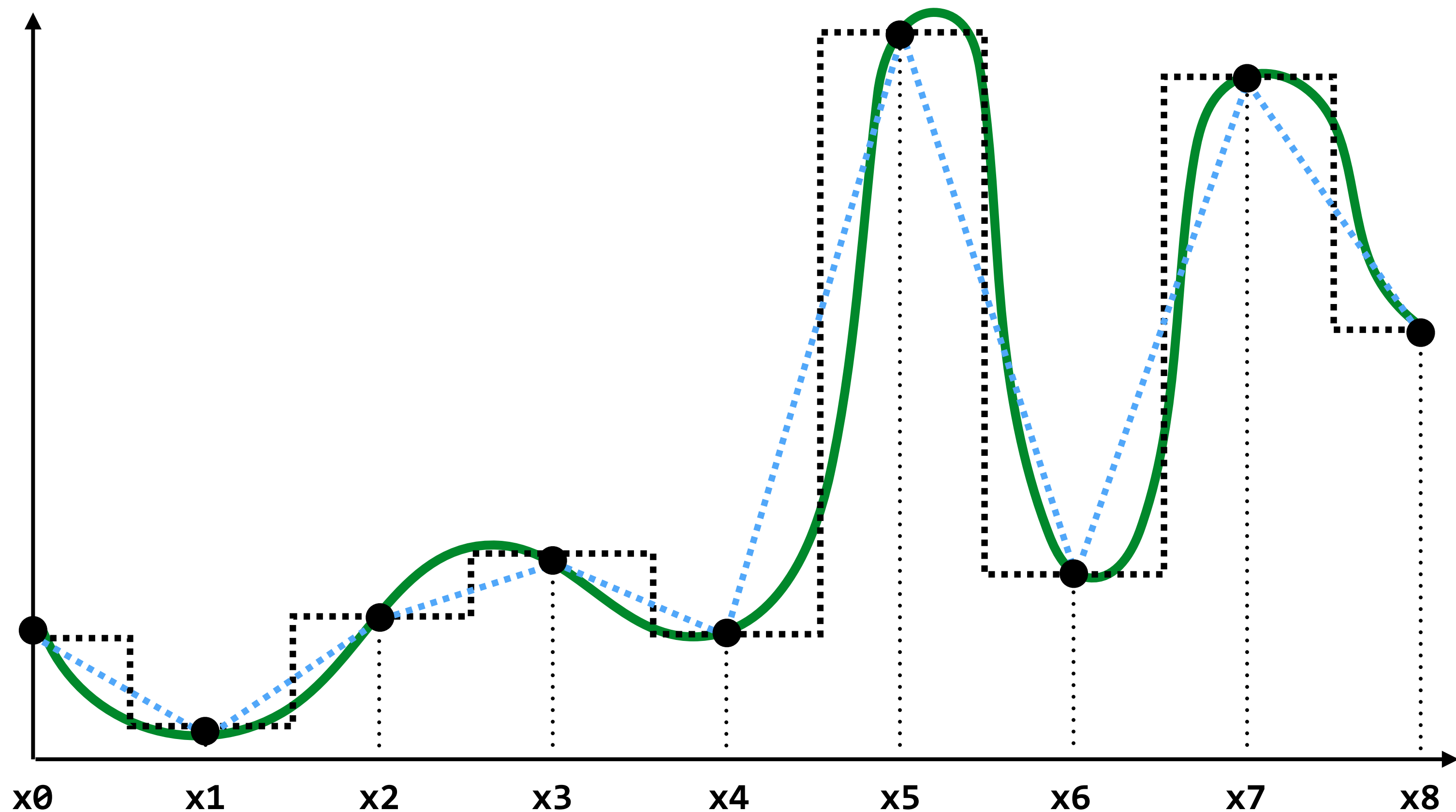
$f_{recon}(x)$  = linear interpolation between values of two closest samples to  $x$



# How can we represent the signal more accurately?



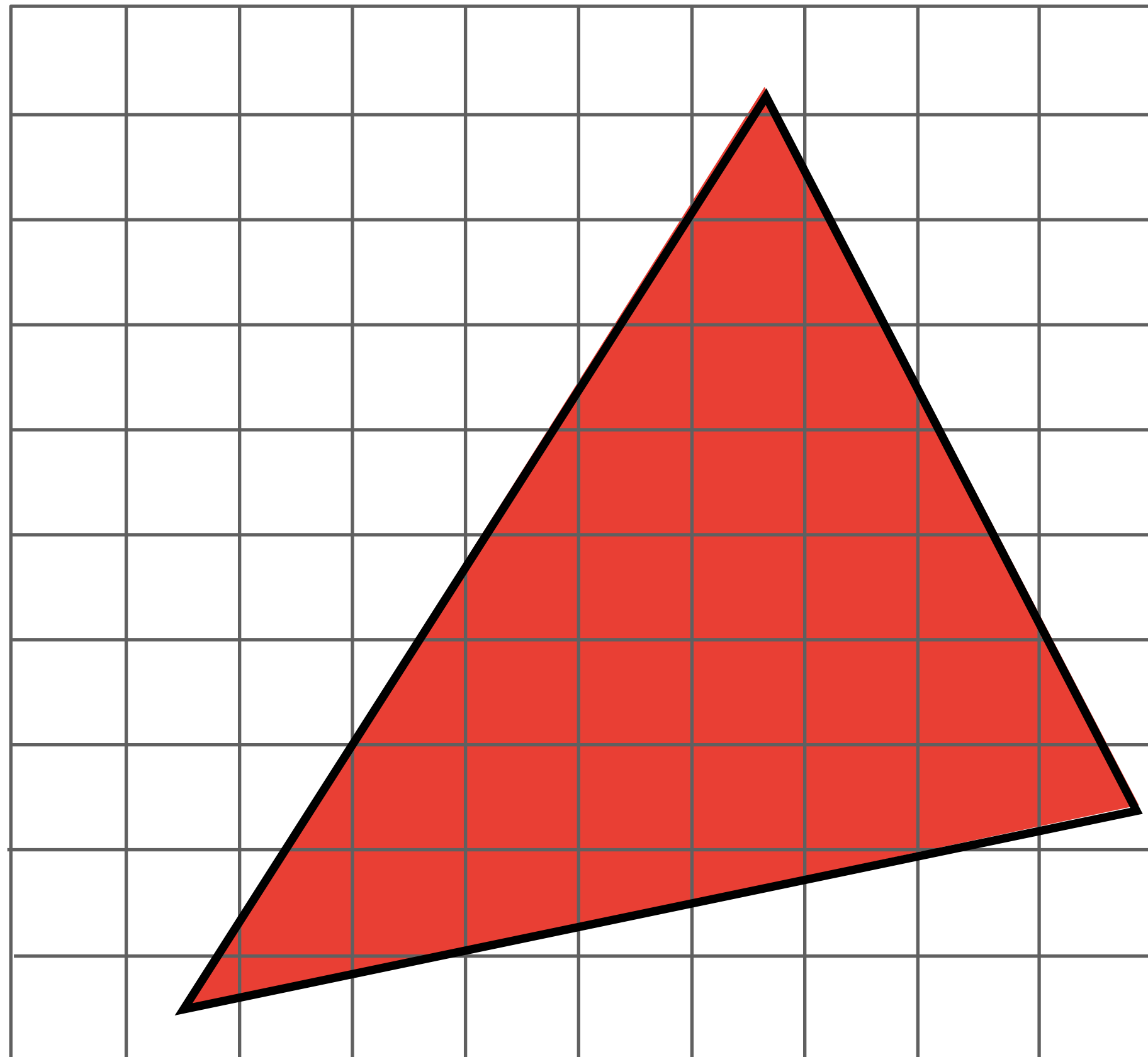
# Reconstructions from denser sampling



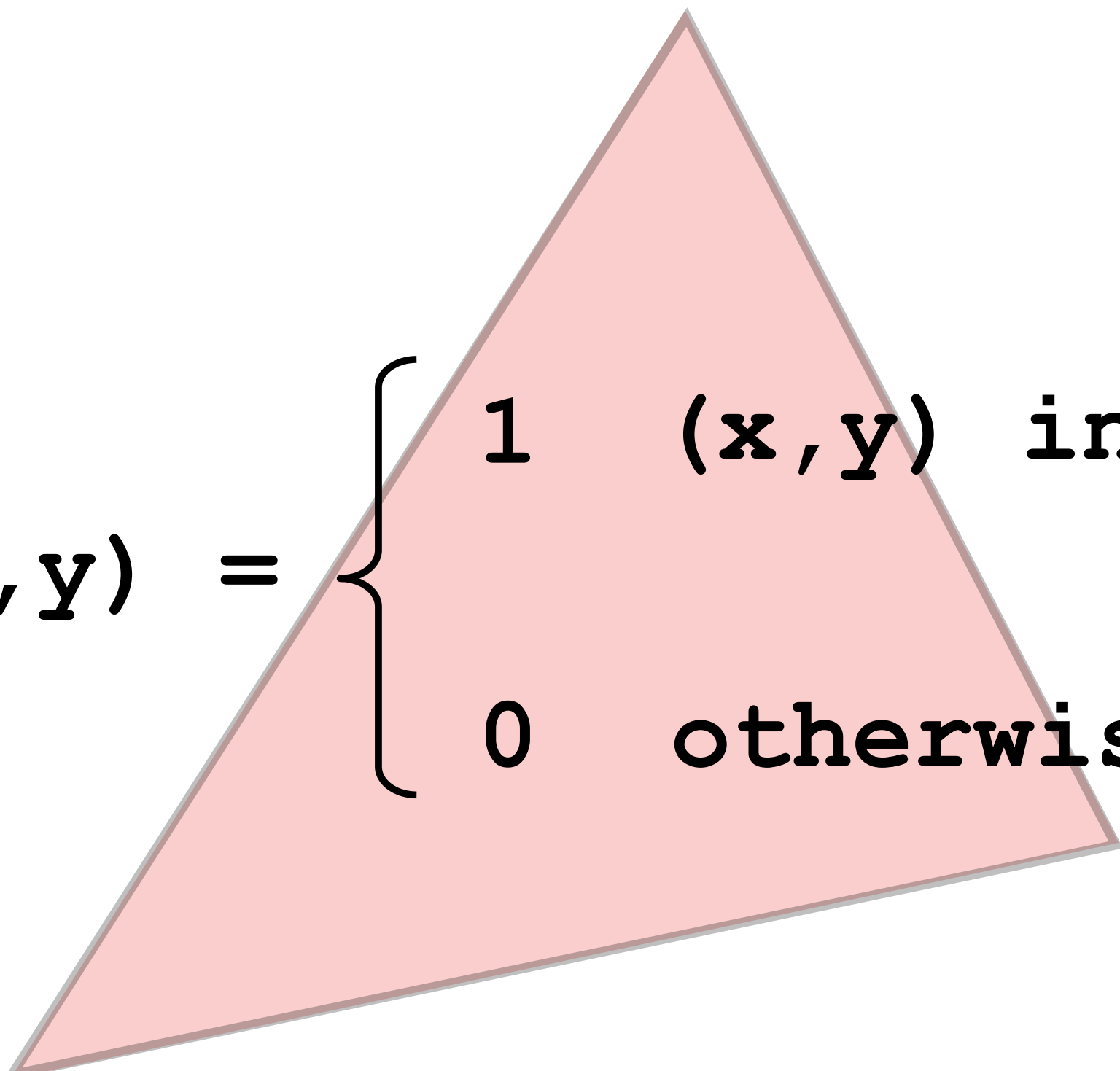
..... = reconstruction via nearest

..... = reconstruction via linear interpolation

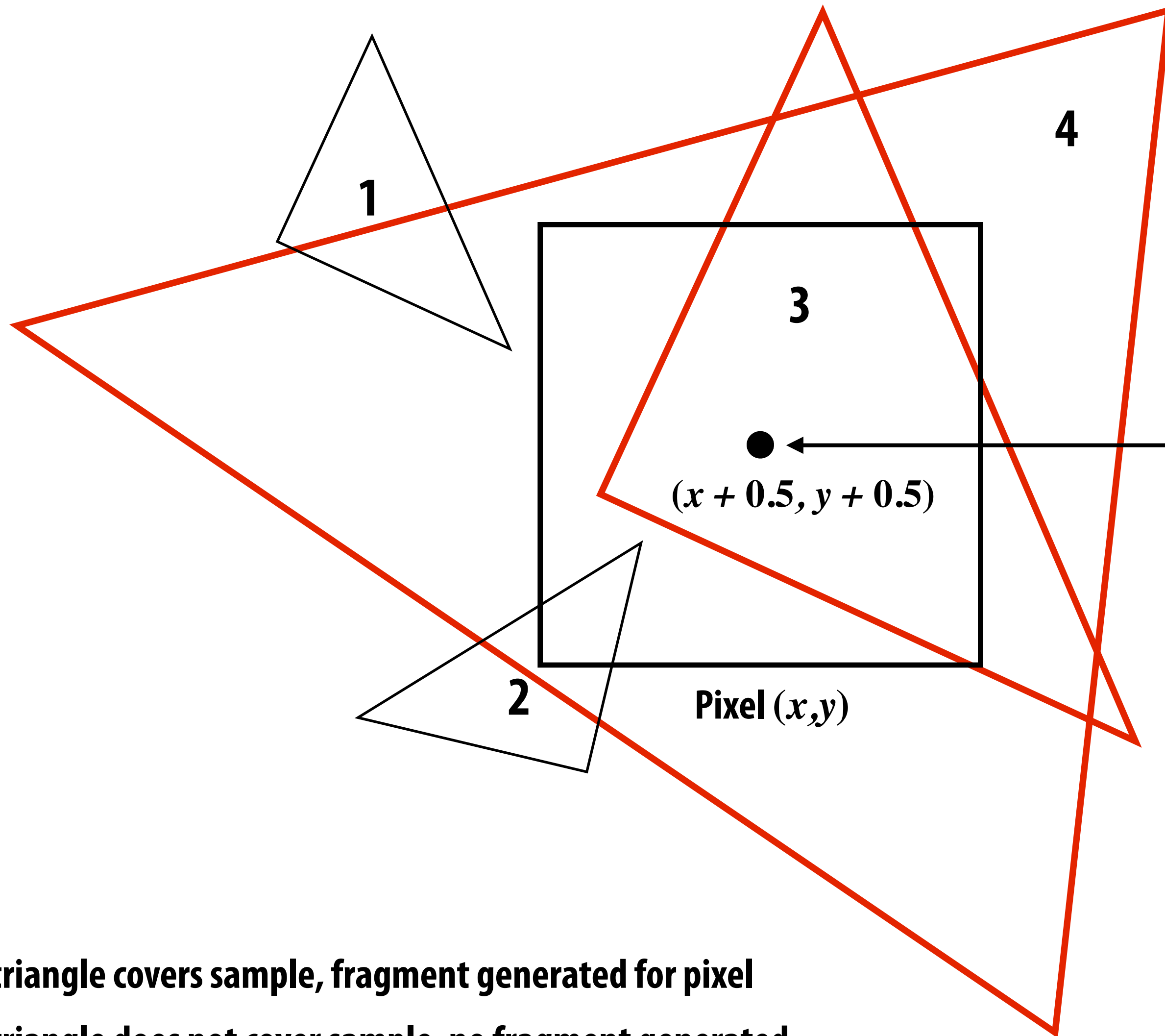
# Drawing a triangle by 2D sampling



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


$$\text{inside}(t, x, y) = \begin{cases} 1 & (x, y) \text{ in triangle } t \\ 0 & \text{otherwise} \end{cases}$$

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

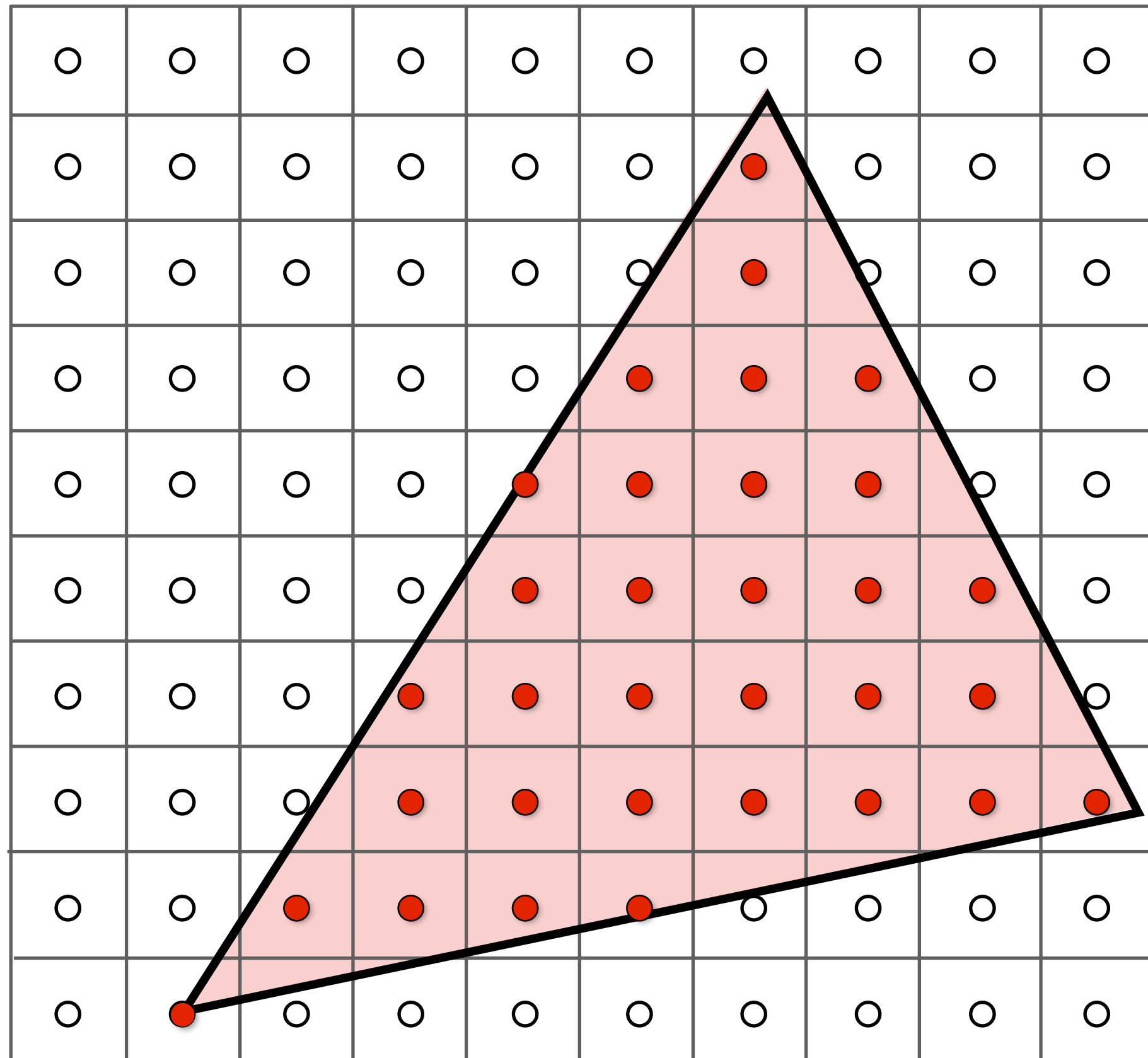


**Example:**  
Here I chose the sample position to be at the pixel center.

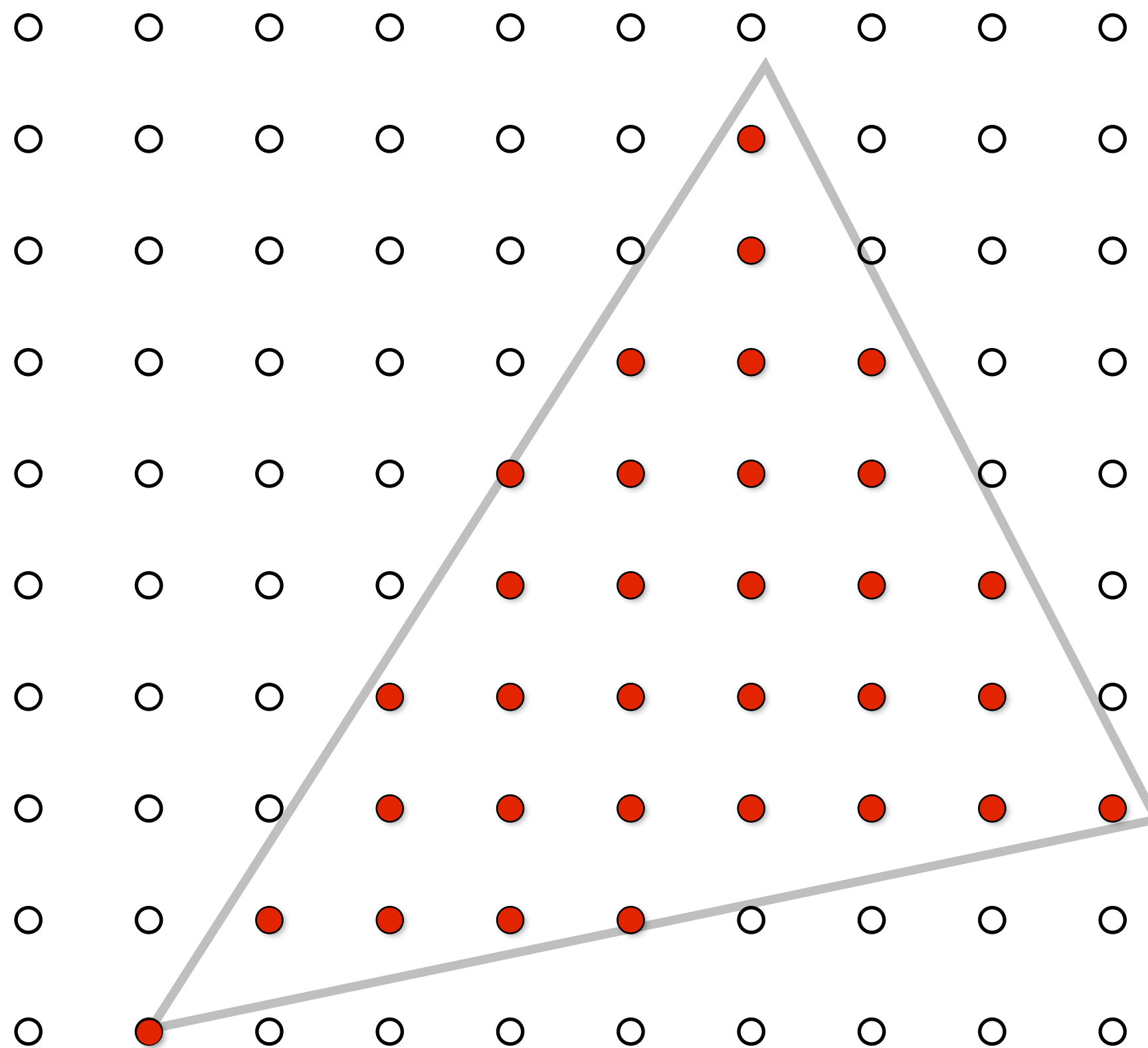
-  = triangle covers sample, fragment generated for pixel
-  = triangle does not cover sample, no fragment generated



# Sample coverage at pixel centers



# Sample coverage at pixel centers



# Rasterization = sampling a 2D indicator 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);
```

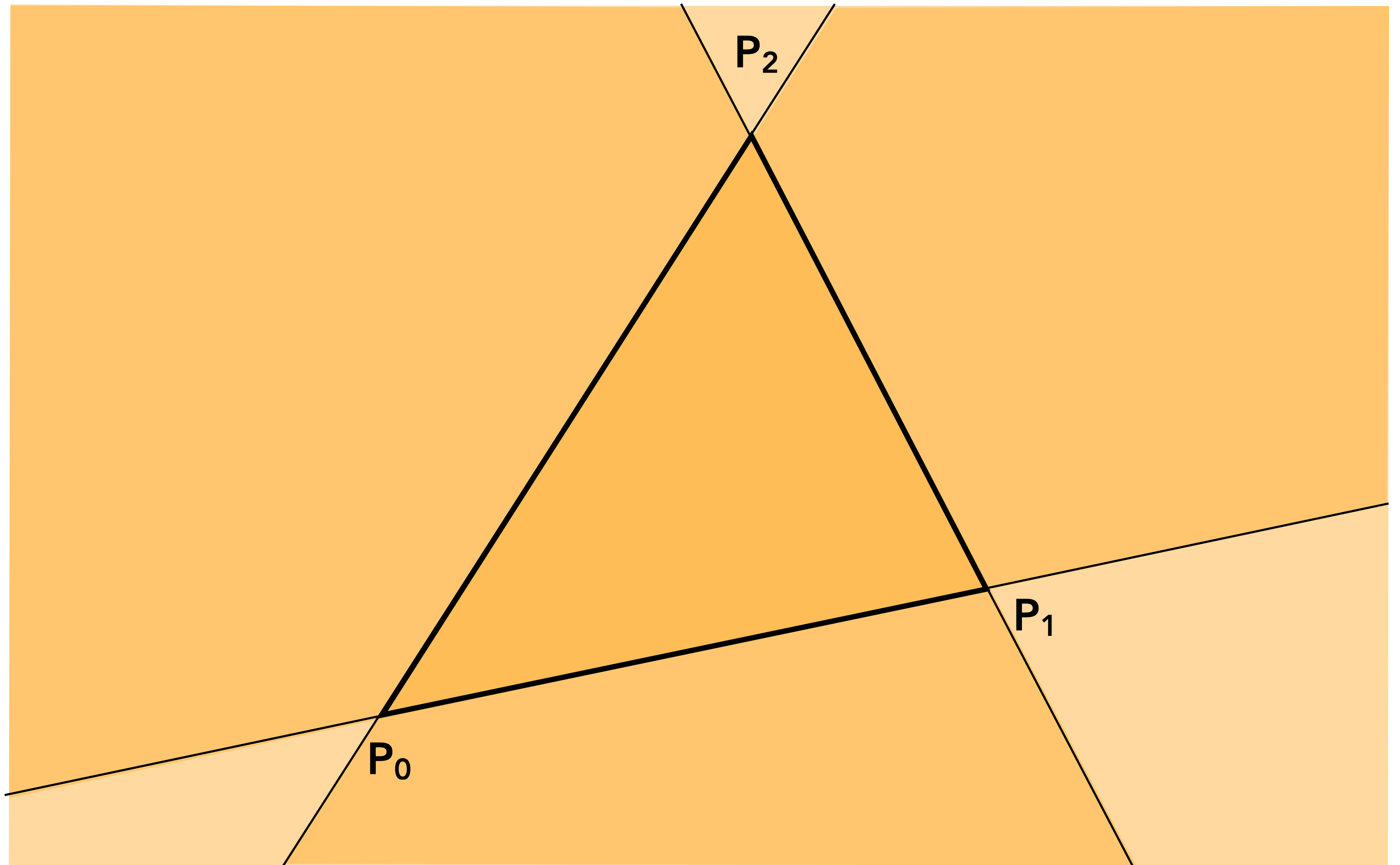
- Rasterize triangle tri by sampling the function

```
f(x, y) = inside(tri, x, y)
```



**Evaluating `inside(tri, x, y)`**

# Triangle = intersection of three half planes



# Each line defines two half-planes

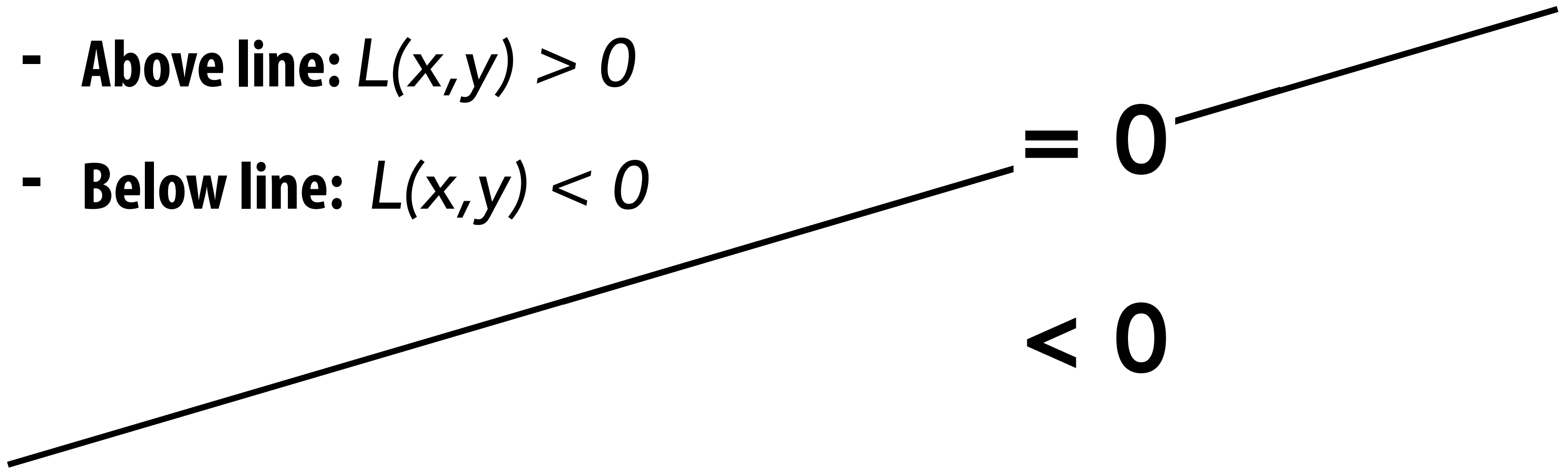
## ■ Implicit line equation

- $L(x,y) = Ax + By + C$

- **On line:**  $L(x,y) = 0$   $\geq 0$

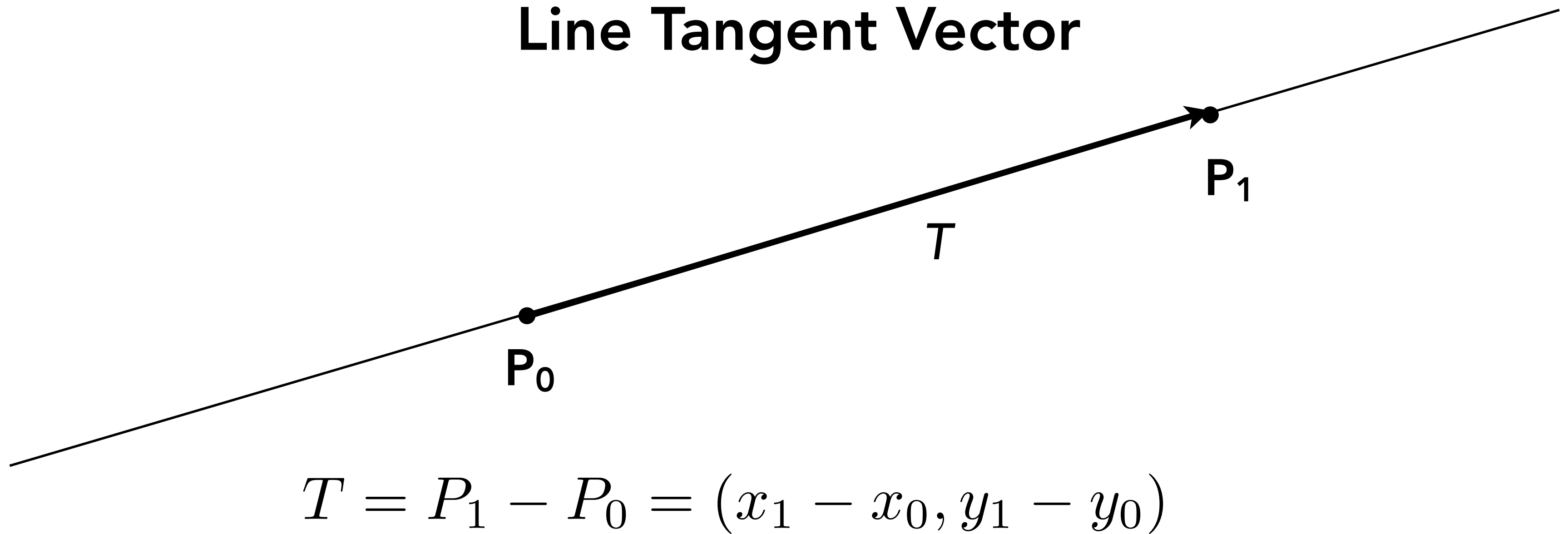
- **Above line:**  $L(x,y) > 0$   $= 0$

- **Below line:**  $L(x,y) < 0$   $< 0$

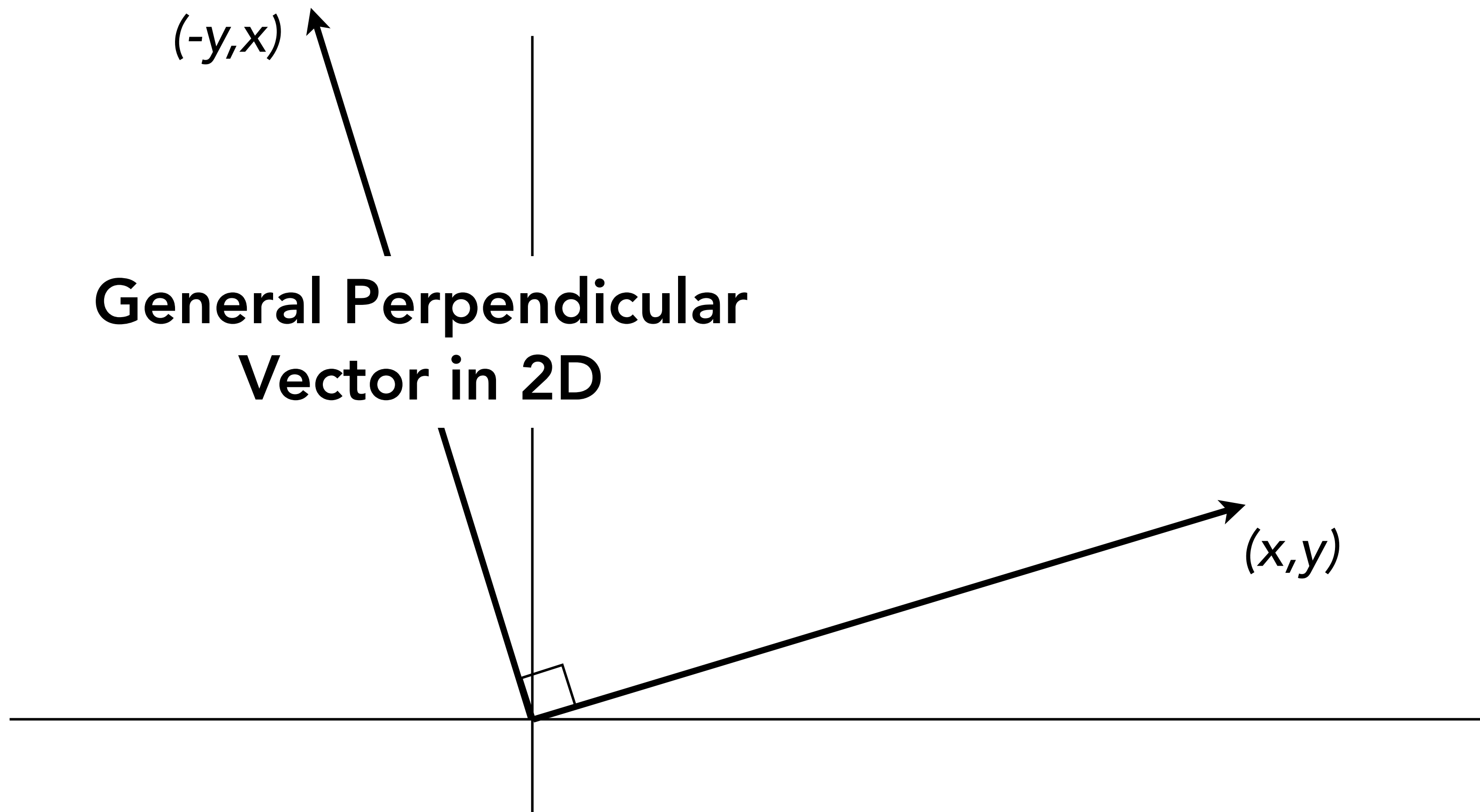


# Line equation derivation

## Line Tangent Vector



# Line equation derivation

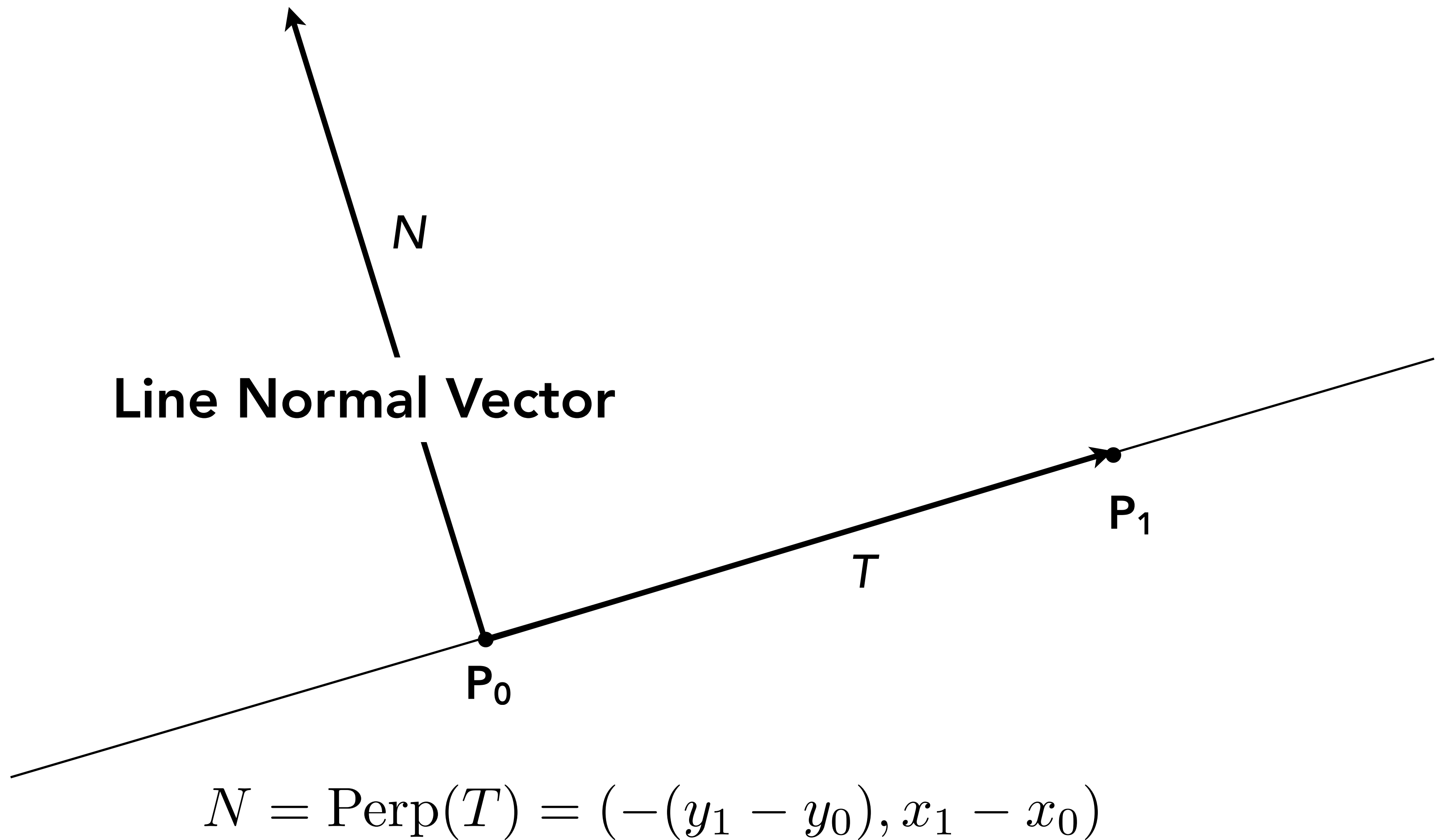


General Perpendicular  
Vector in 2D

$$\text{Perp}(x, y) = (-y, x)$$

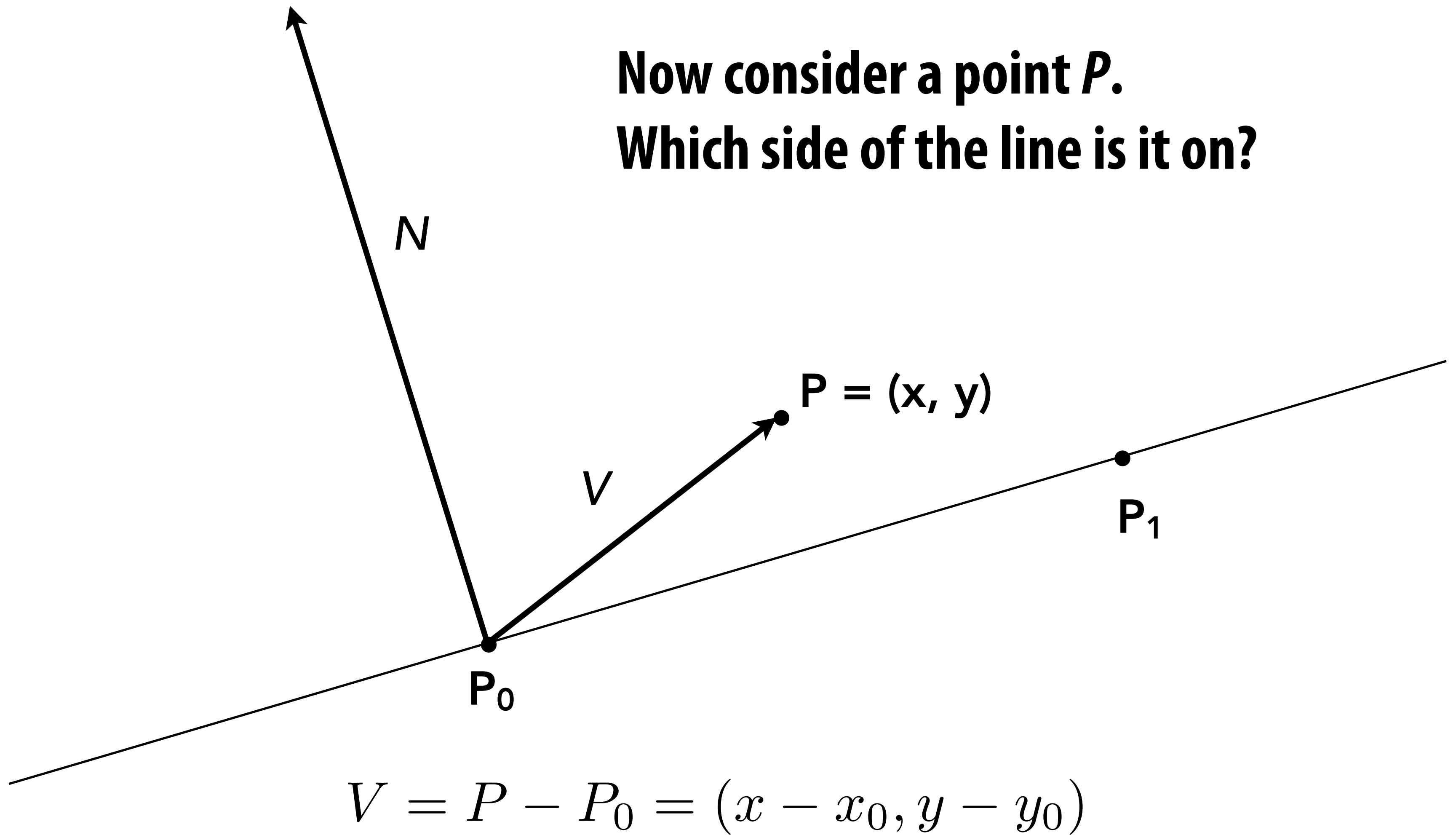


# Line equation derivation

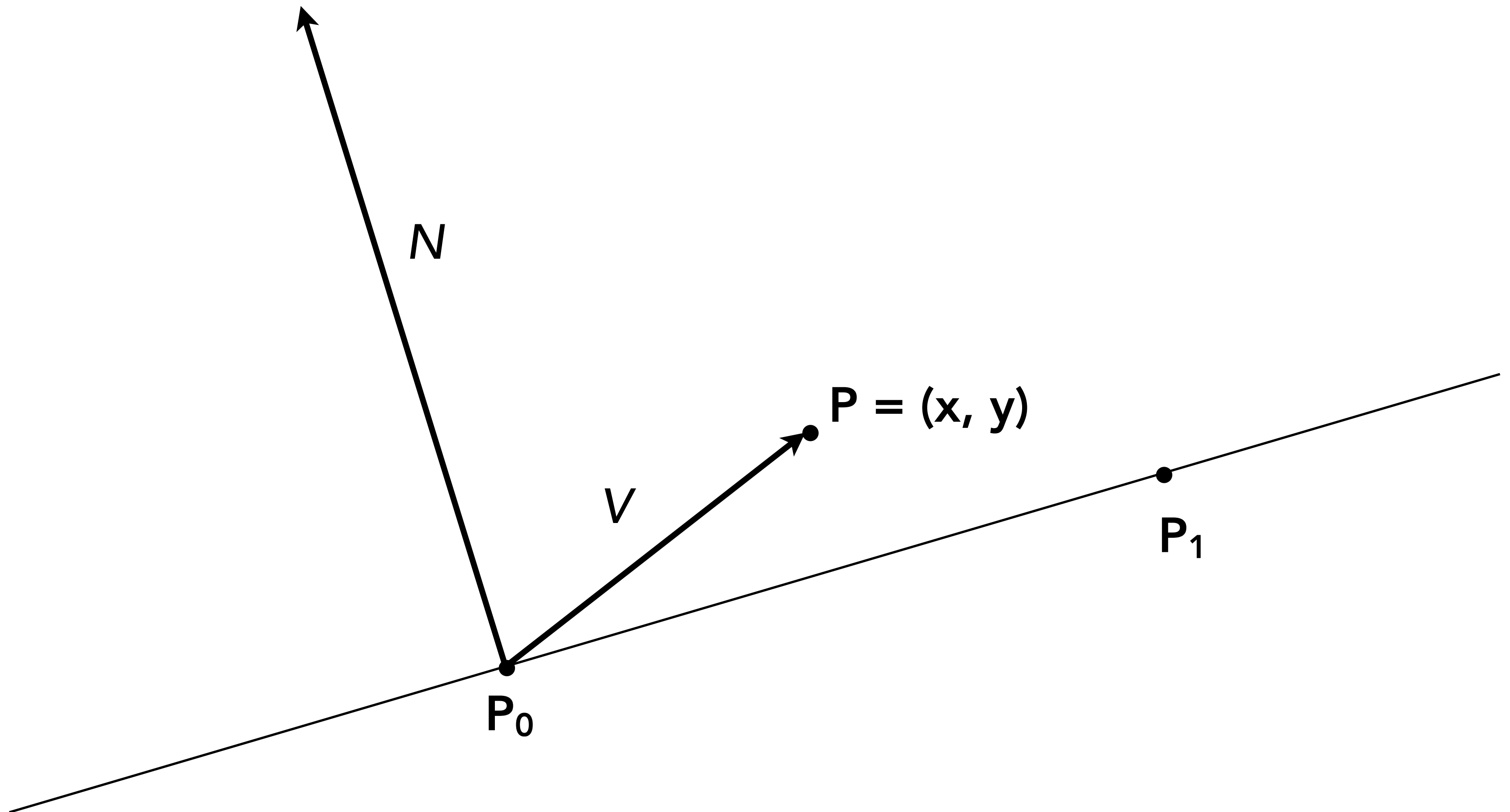


# Line equation derivation

Now consider a point  $P$ .  
Which side of the line is it on?

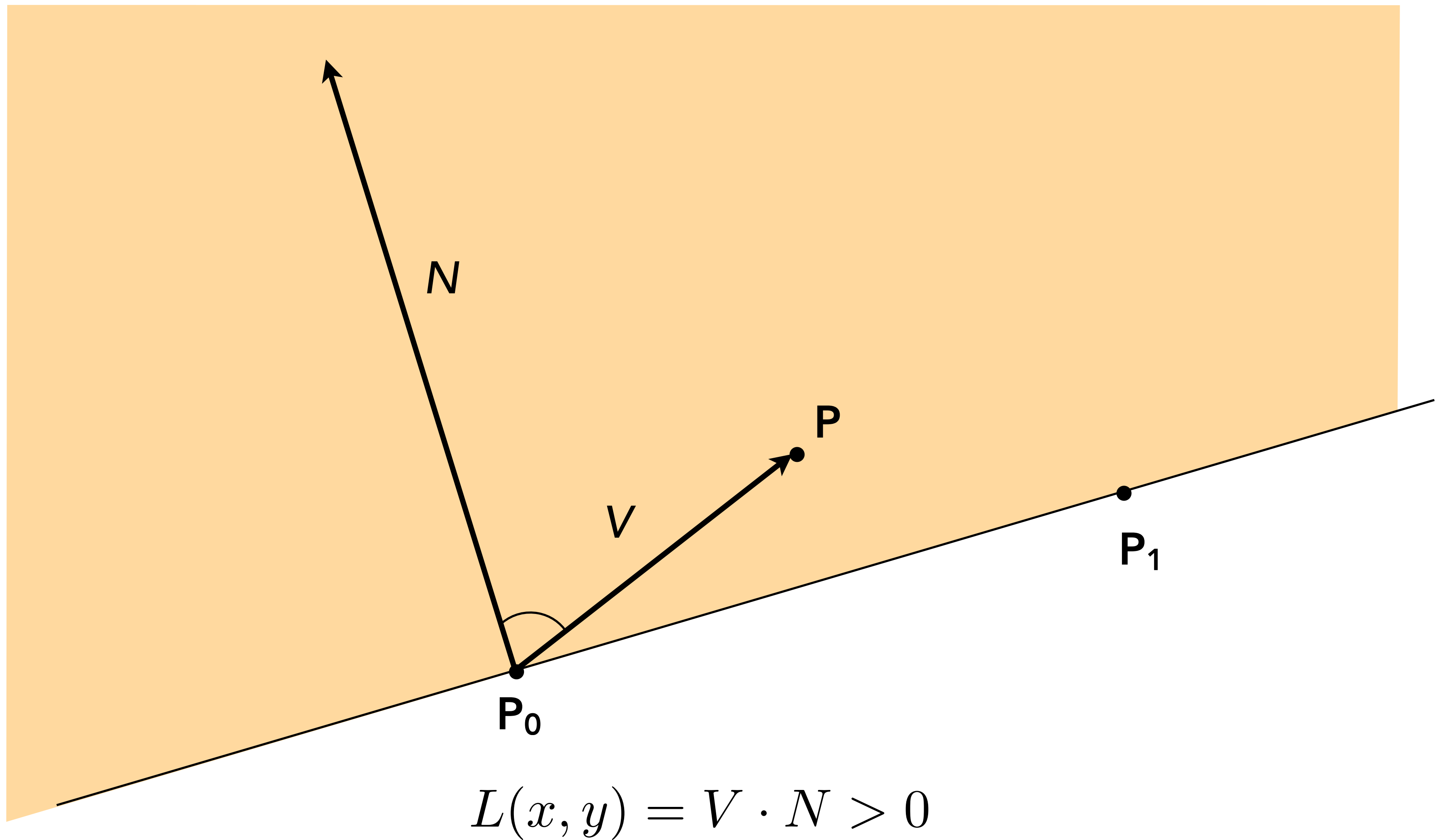


# Line equation derivation

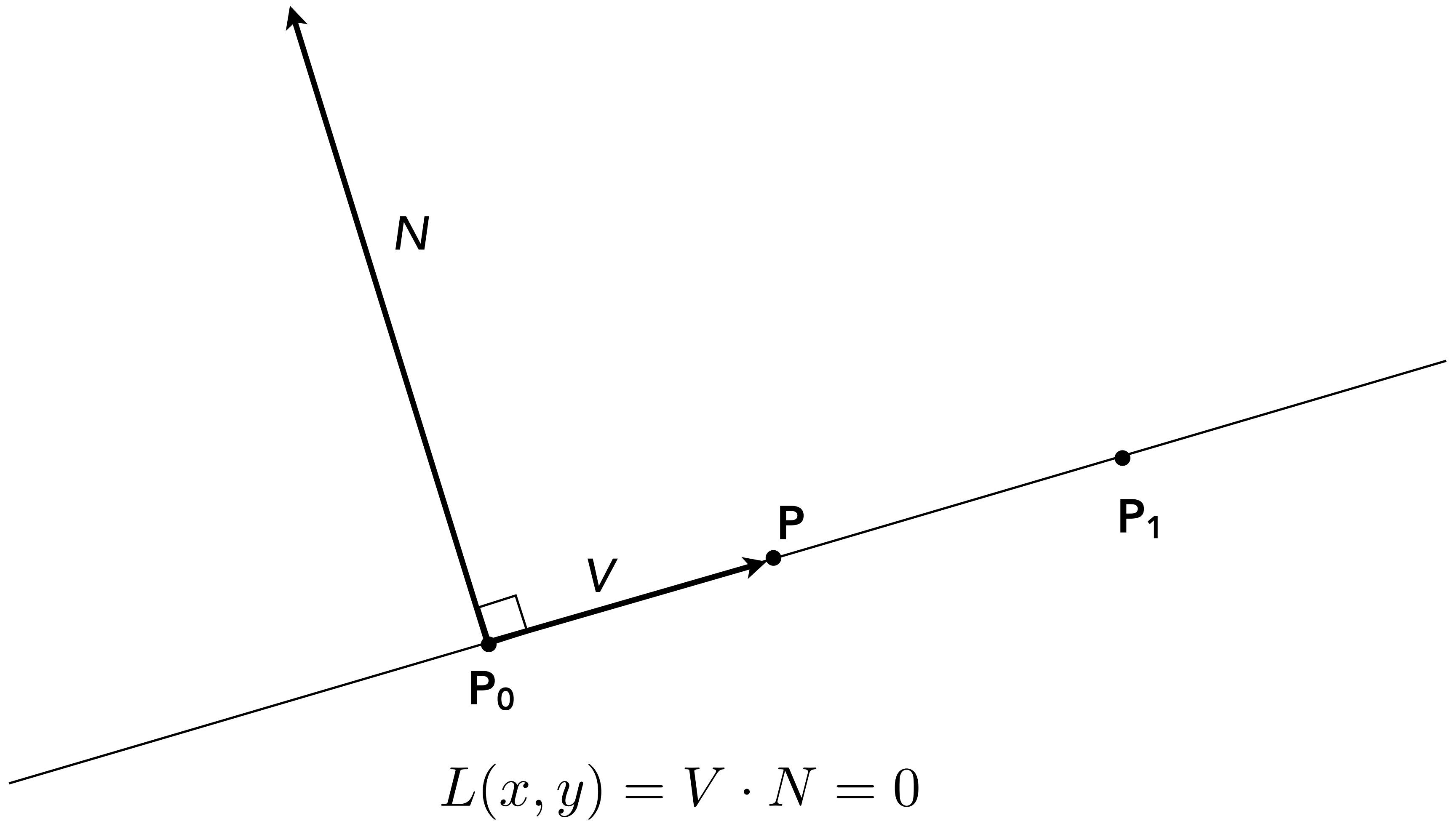


$$L(x, y) = V \cdot N = -(x - x_0)(y_1 - y_0) + (y - y_0)(x_1 - x_0)$$

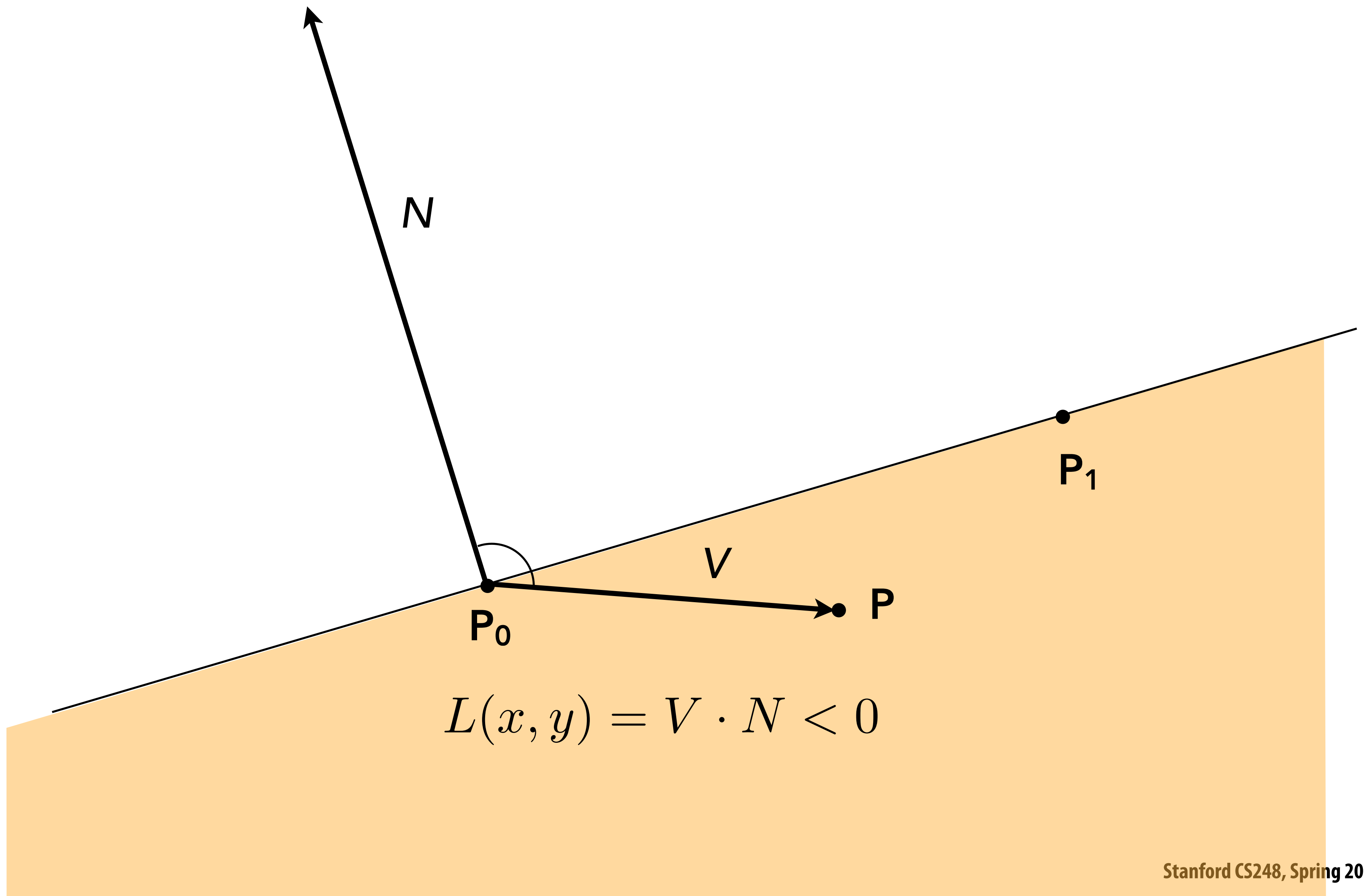
# Line equation tests



# Line equation tests



# Line equation tests



# Point-in-triangle test

$$P_i = (X_i, Y_i)$$

$$dX_i = X_{i+1} - X_i$$

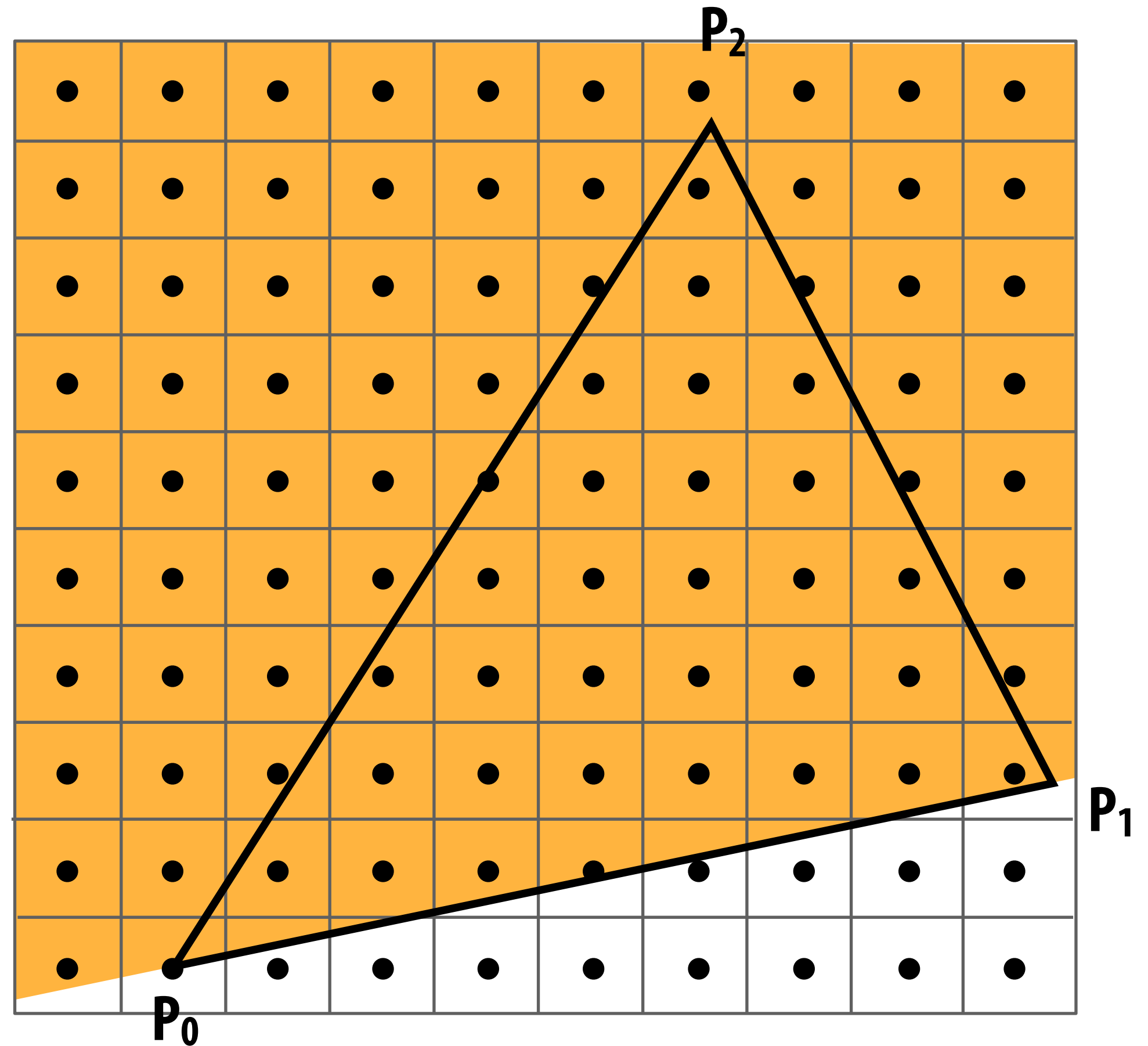
$$dY_i = Y_{i+1} - Y_i$$

$$\begin{aligned} L_i(x, y) &= (x - X_i) dY_i - (y - Y_i) dX_i \\ &= A_i x + B_i y + C_i \end{aligned}$$

$L_i(x, y) = 0$  : point on edge

$> 0$  : outside edge

$< 0$  : inside edge



$$L_0(x, y) > 0$$

# Point-in-triangle test

$$P_i = (X_i, Y_i)$$

$$dX_i = X_{i+1} - X_i$$

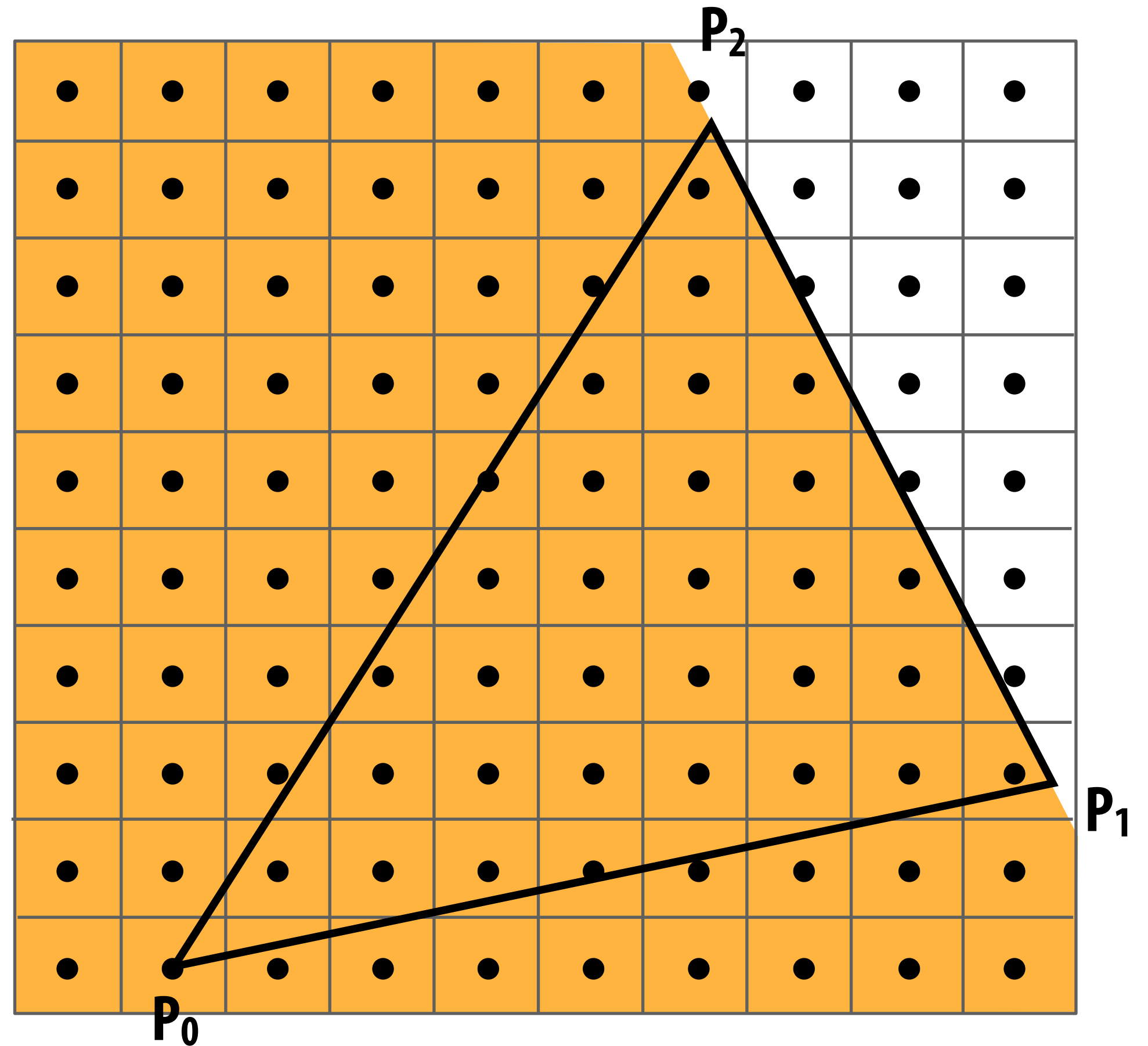
$$dY_i = Y_{i+1} - Y_i$$

$$\begin{aligned} L_i(x, y) &= (x - X_i) dY_i - (y - Y_i) dX_i \\ &= A_i x + B_i y + C_i \end{aligned}$$

$L_i(x, y) = 0$  : point on edge

$> 0$  : outside edge

$< 0$  : inside edge



$$L_1(x, y) > 0$$



# Point-in-triangle test

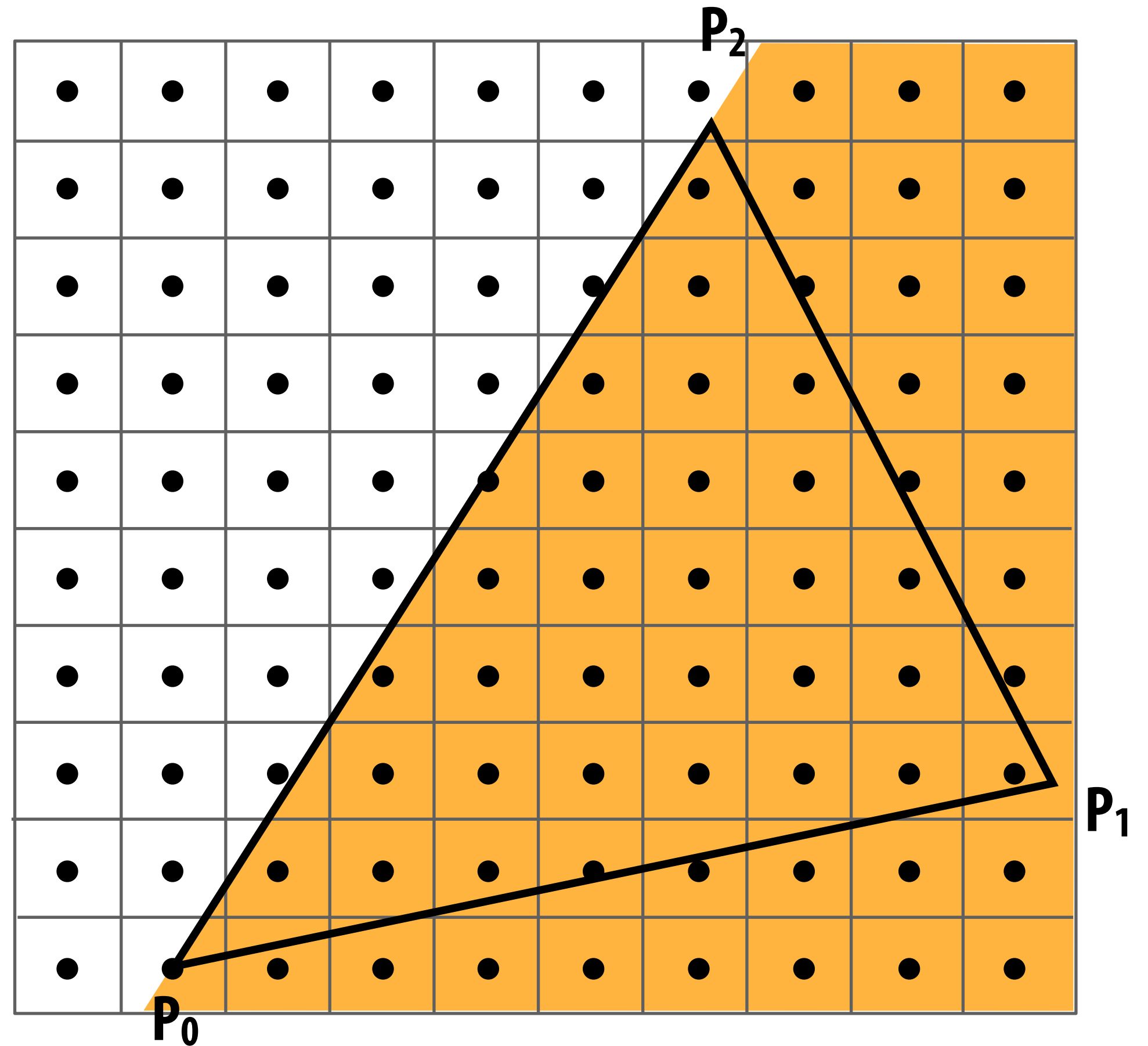
$$P_i = (X_i, Y_i)$$

$$dX_i = X_{i+1} - X_i$$

$$dY_i = Y_{i+1} - Y_i$$

$$\begin{aligned} L_i(x, y) &= (x - X_i) dY_i - (y - Y_i) dX_i \\ &= A_i x + B_i y + C_i \end{aligned}$$

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



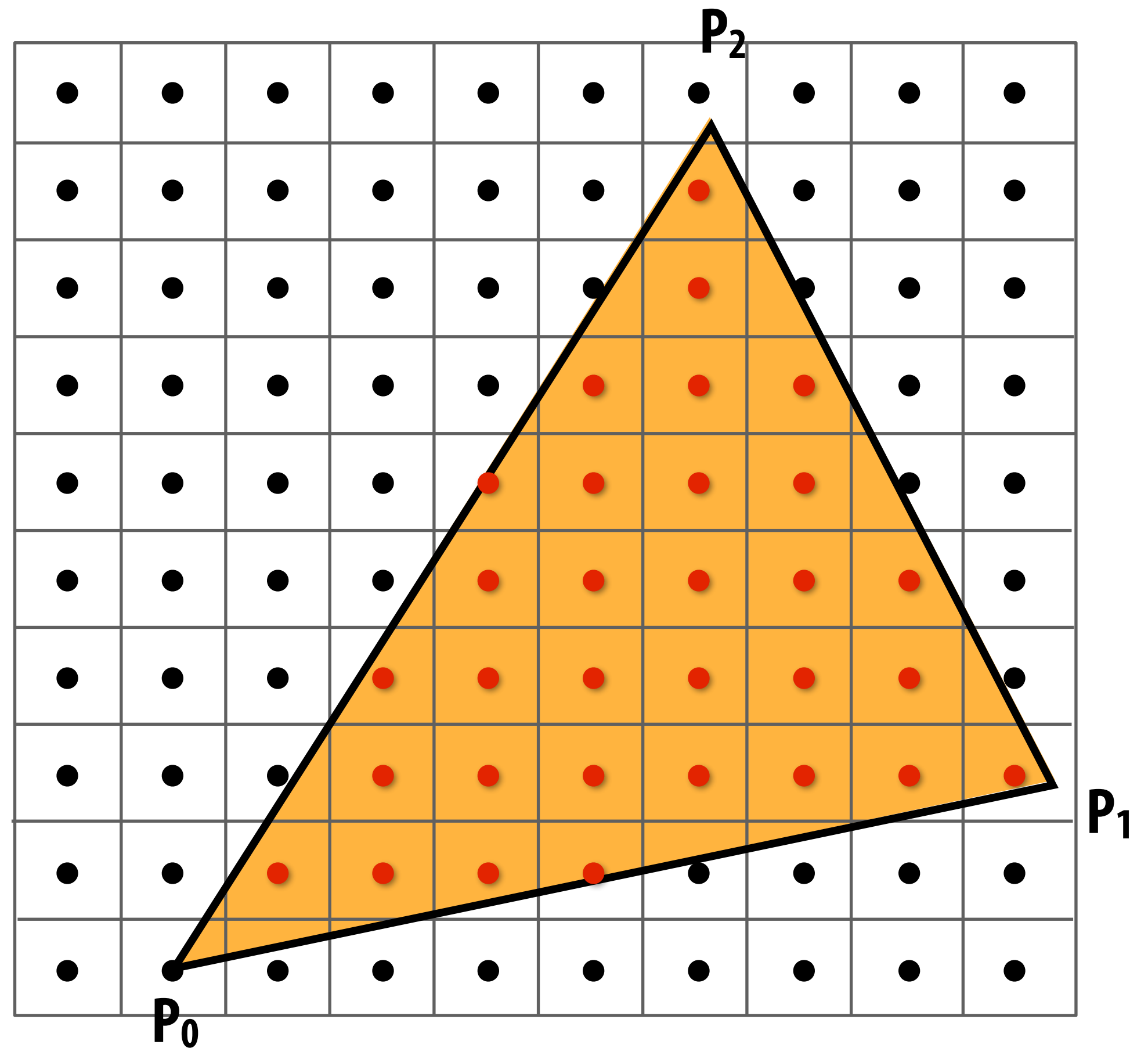
$$L_2(x, y) > 0$$

# Point-in-triangle test

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

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

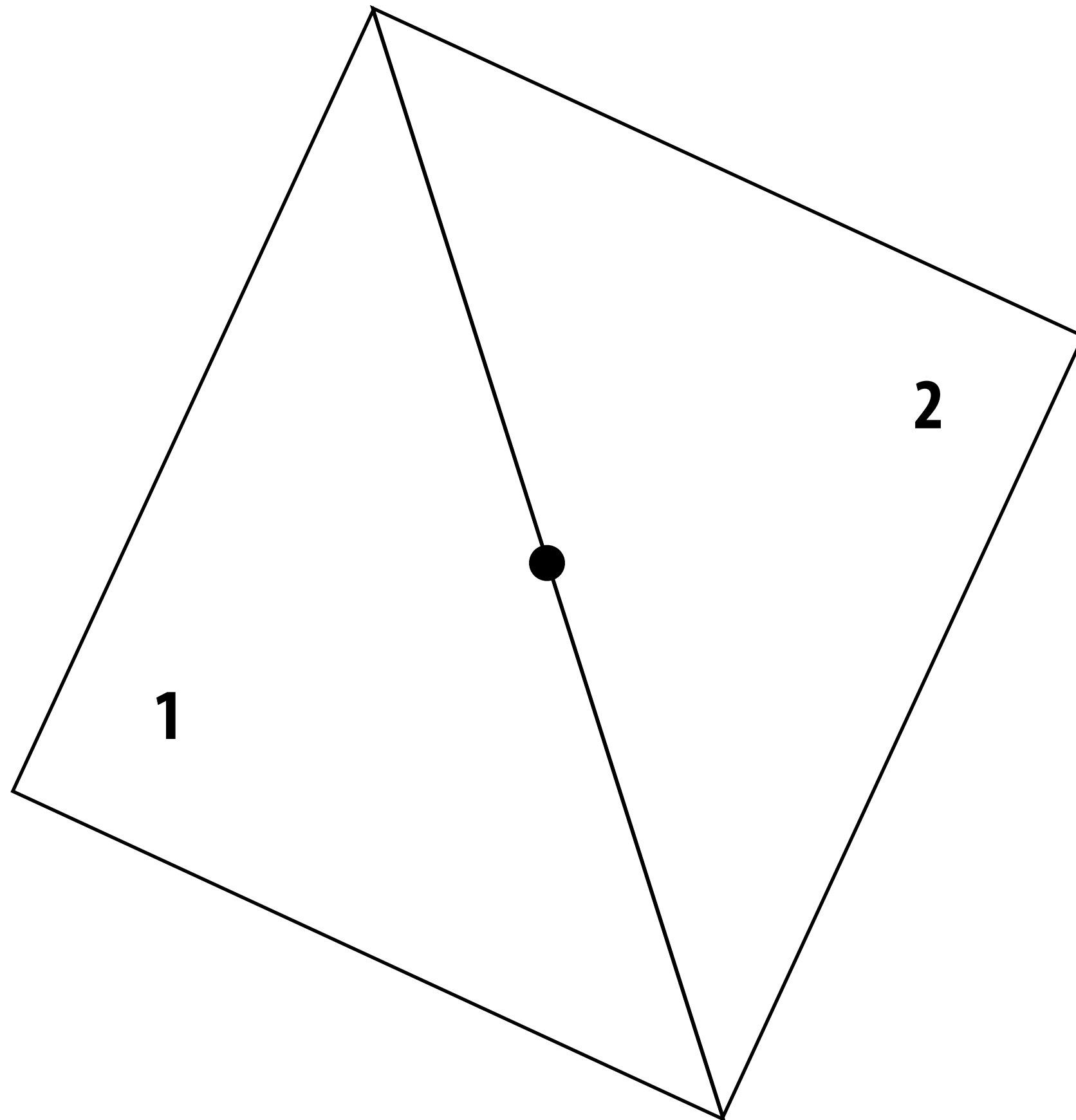
**Note: actual implementation of  $inside(sx, sy)$  involves  $\leq$  checks based on the triangle coverage edge rules (see next slides)**



Sample points inside triangle are highlighted red.

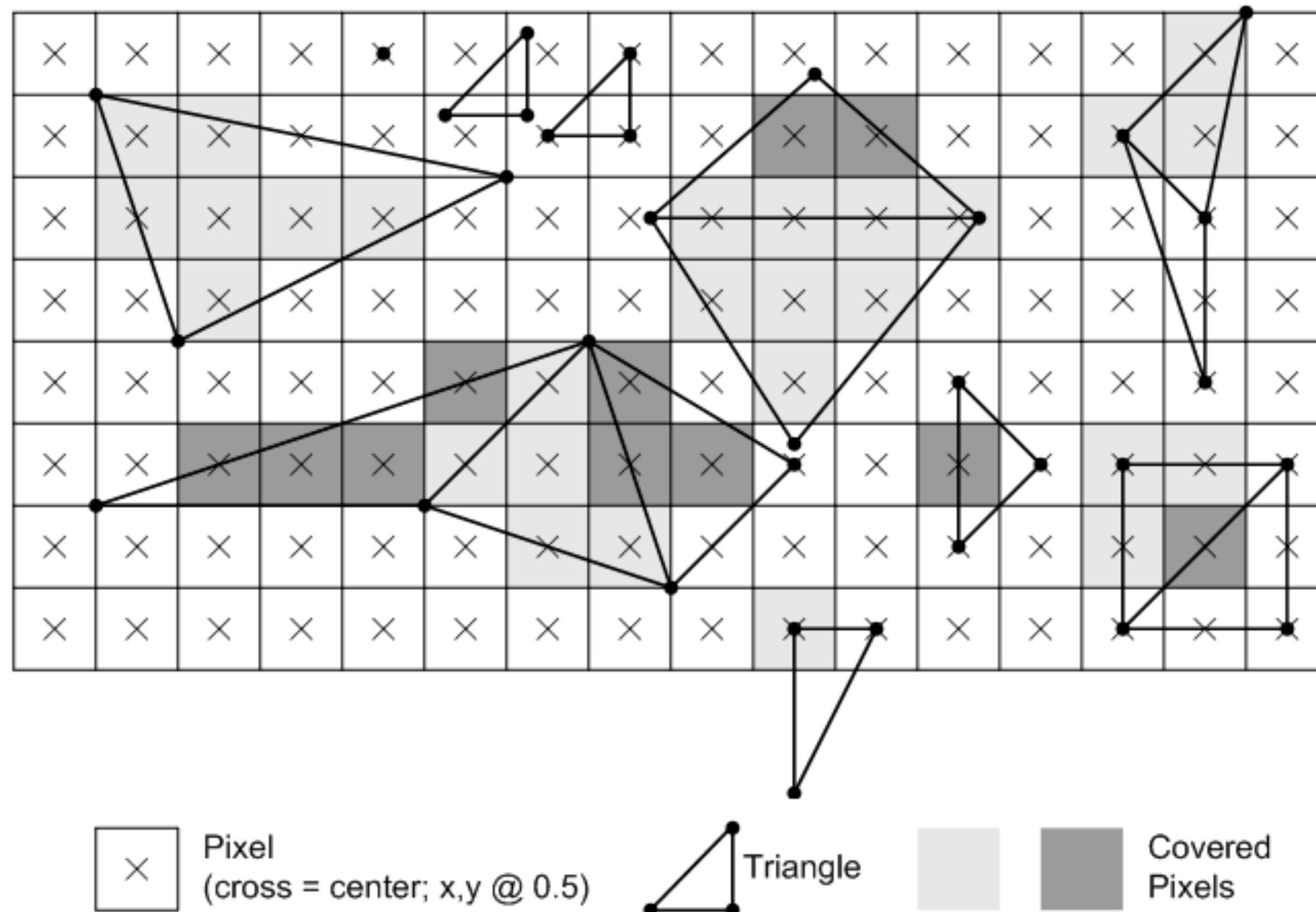
# Edge cases (literally)

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

$$P_i = (X_i, Y_i)$$

$$dX_i = X_{i+1} - X_i$$

$$dY_i = Y_{i+1} - Y_i$$

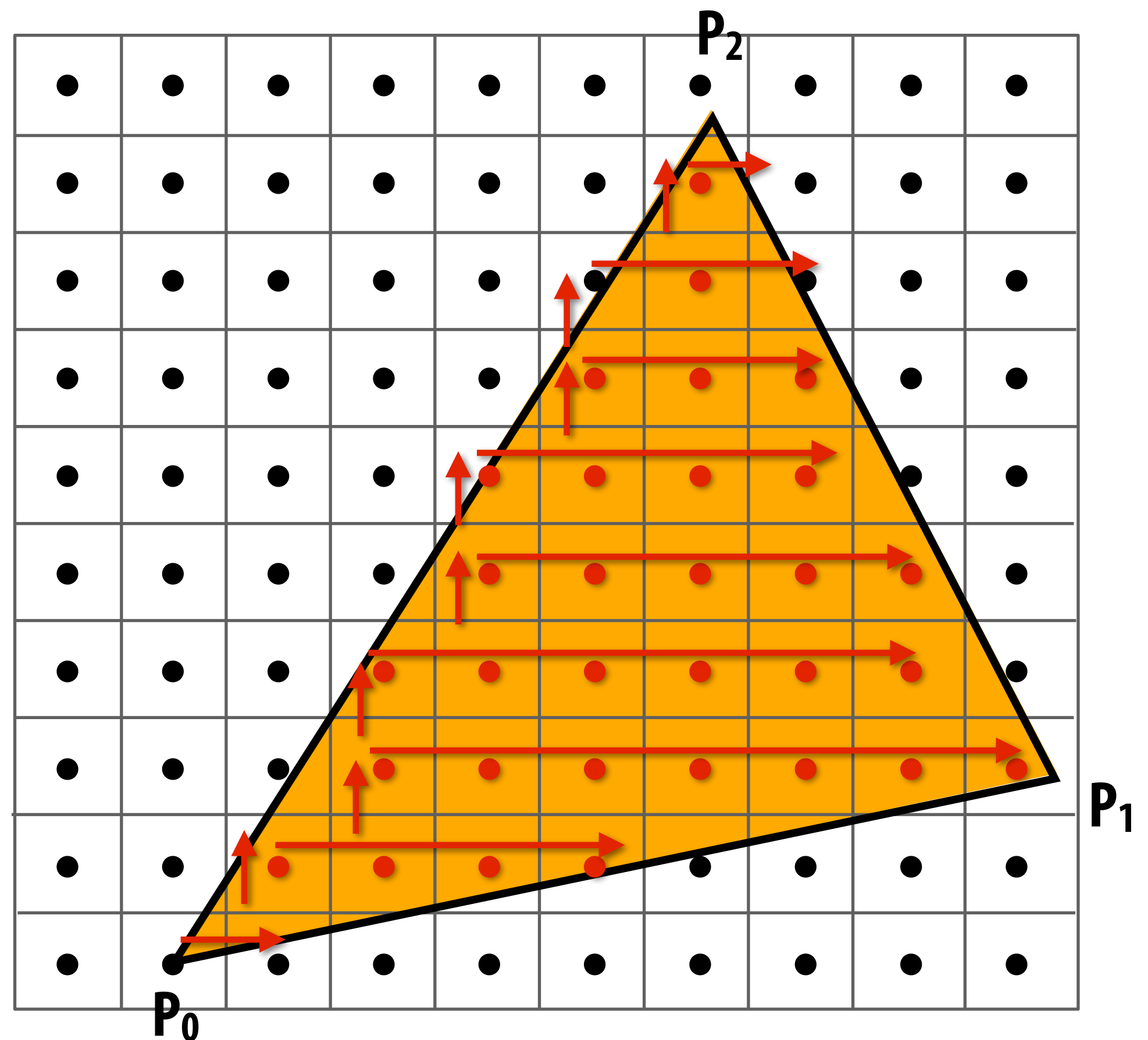
$$\begin{aligned} L_i(x, y) &= (x - X_i) dY_i - (y - Y_i) dX_i \\ &= A_i x + B_i y + C_i \end{aligned}$$

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

**Efficient incremental update:**

$$dL_i(x+1, y) = L_i(x, y) + dY_i = L_i(x, y) + A_i$$

$$dL_i(x, y+1) = L_i(x, y) + dX_i = L_i(x, y) + B_i$$



**Incremental update saves computation:**

**Only one addition per edge, per sample test**

**Many traversal orders are possible: backtrack, zig-zag, Hilbert/Morton curves (locality maximizing)**

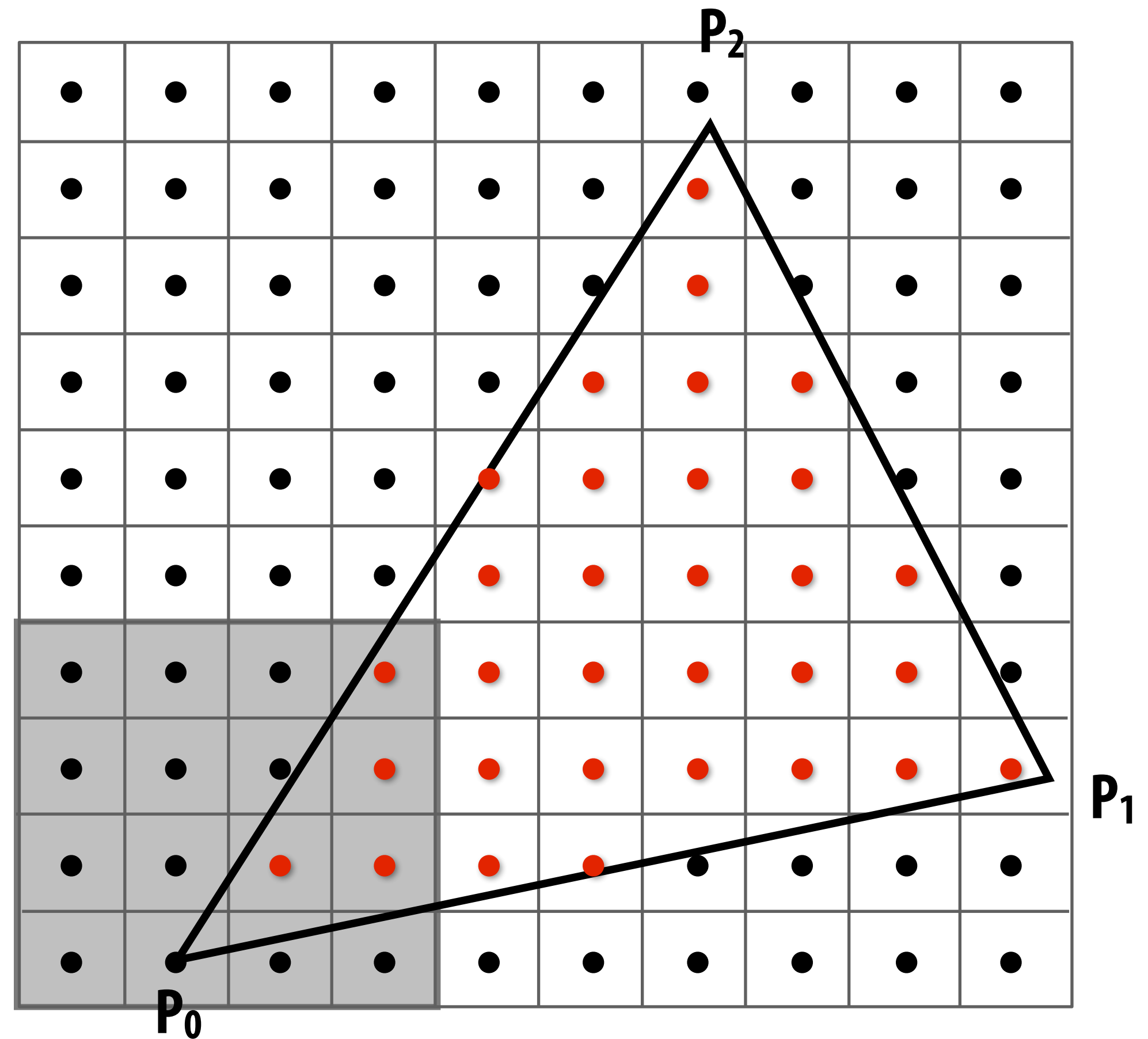
# Modern approach: tiled triangle traversal

Traverse triangle in blocks

Test all samples in block against triangle in parallel

Advantages:

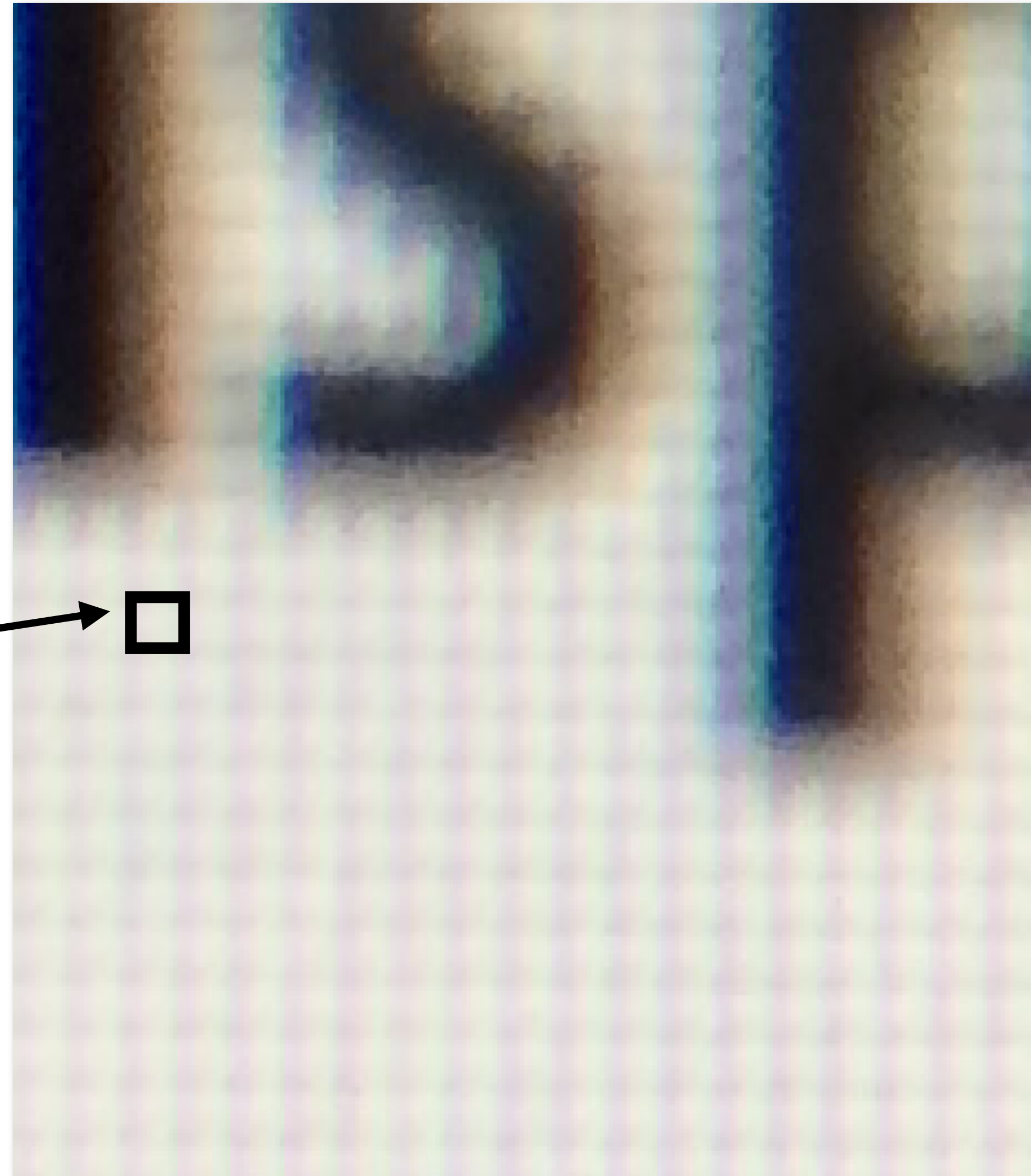
- Simplicity of parallel execution overcomes cost of extra point-in-triangle tests (most triangles cover many samples)
- Can skip sample testing work: entire block not in triangle ("early out"), entire block entirely within triangle ("early in")
- Additional advantage related to accelerating occlusion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



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

# Recall: pixels on a screen

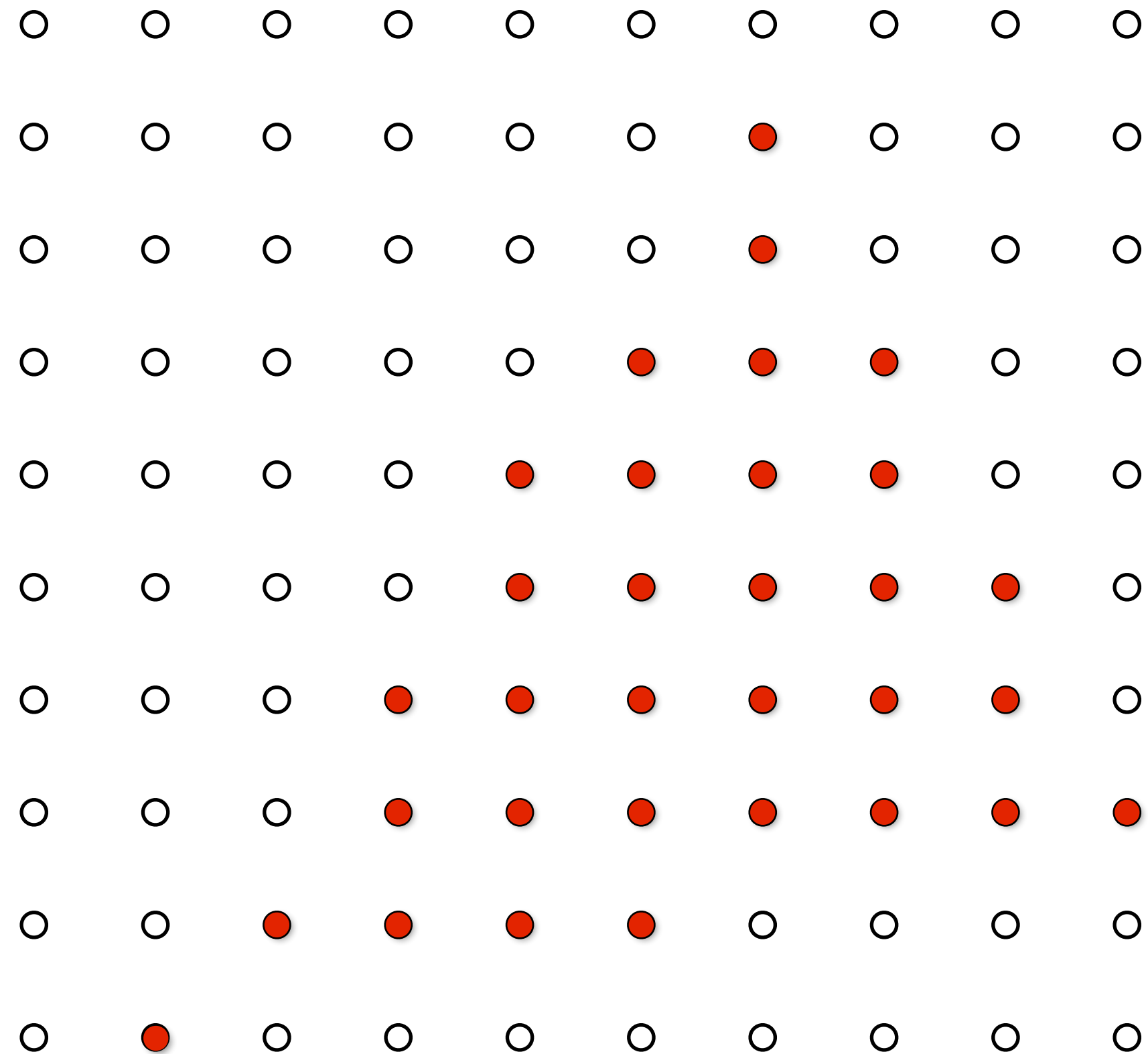
**Each image sample sent to the display is converted into a little square of light of the appropriate color:  
(a pixel = picture element)**



**LCD display  
pixel on my  
laptop**

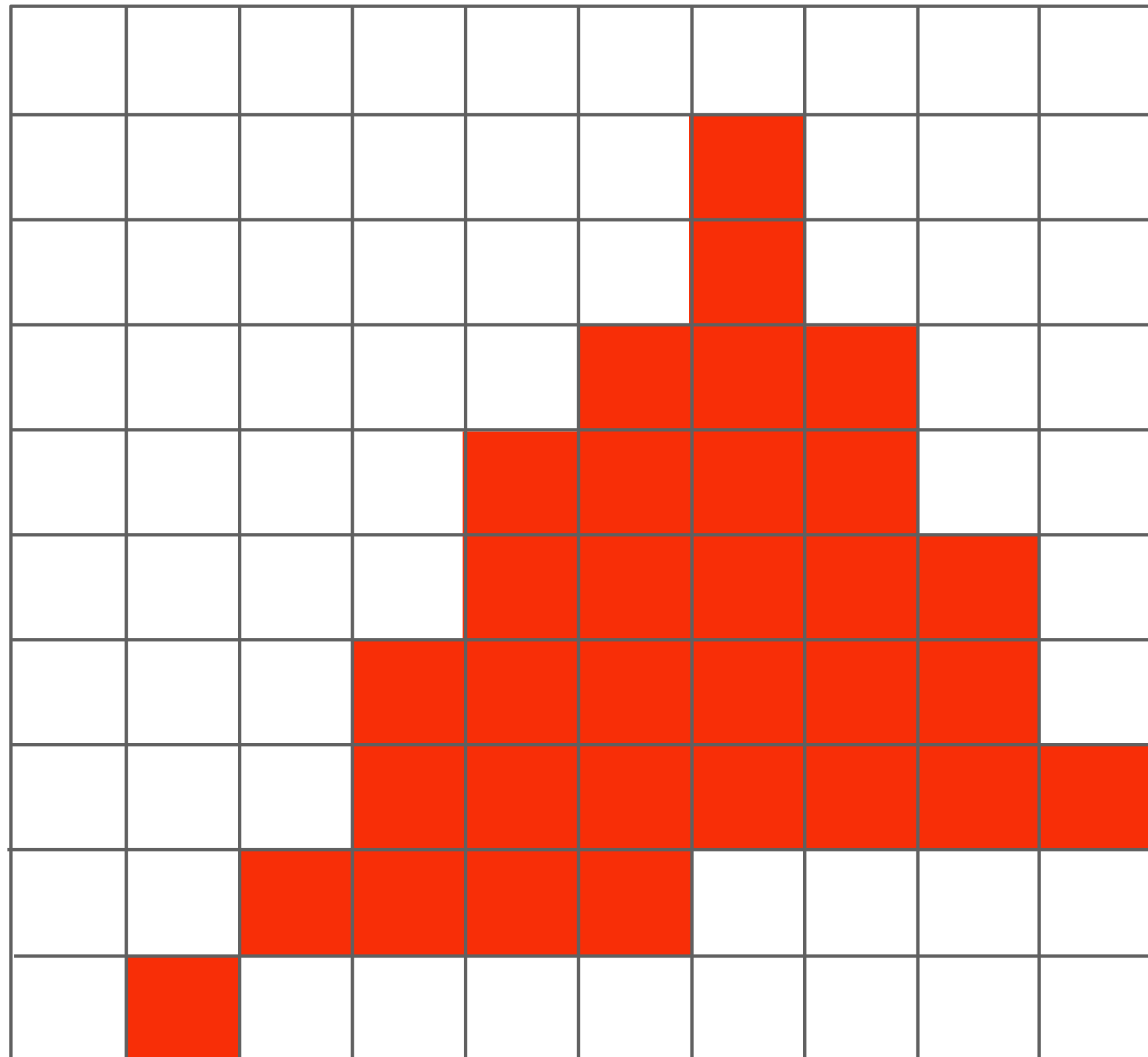
**\* Thinking of each LCD pixel as emitting a square of uniform intensity light of a single color is a bit of an approximation to how real displays work, but it will do for now.**

# So, if we send the display this sampled signal



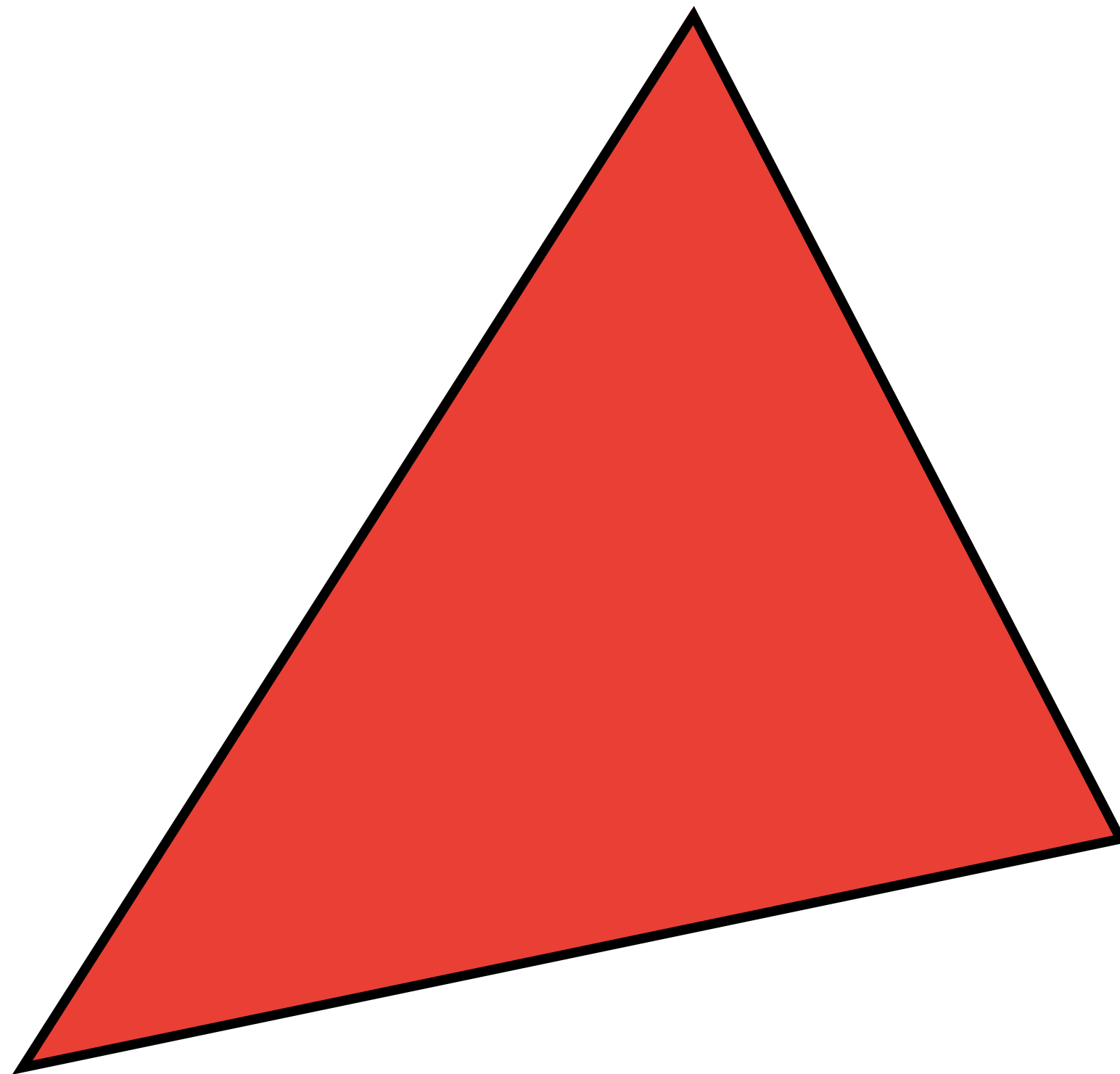


# The display physically emits this signal

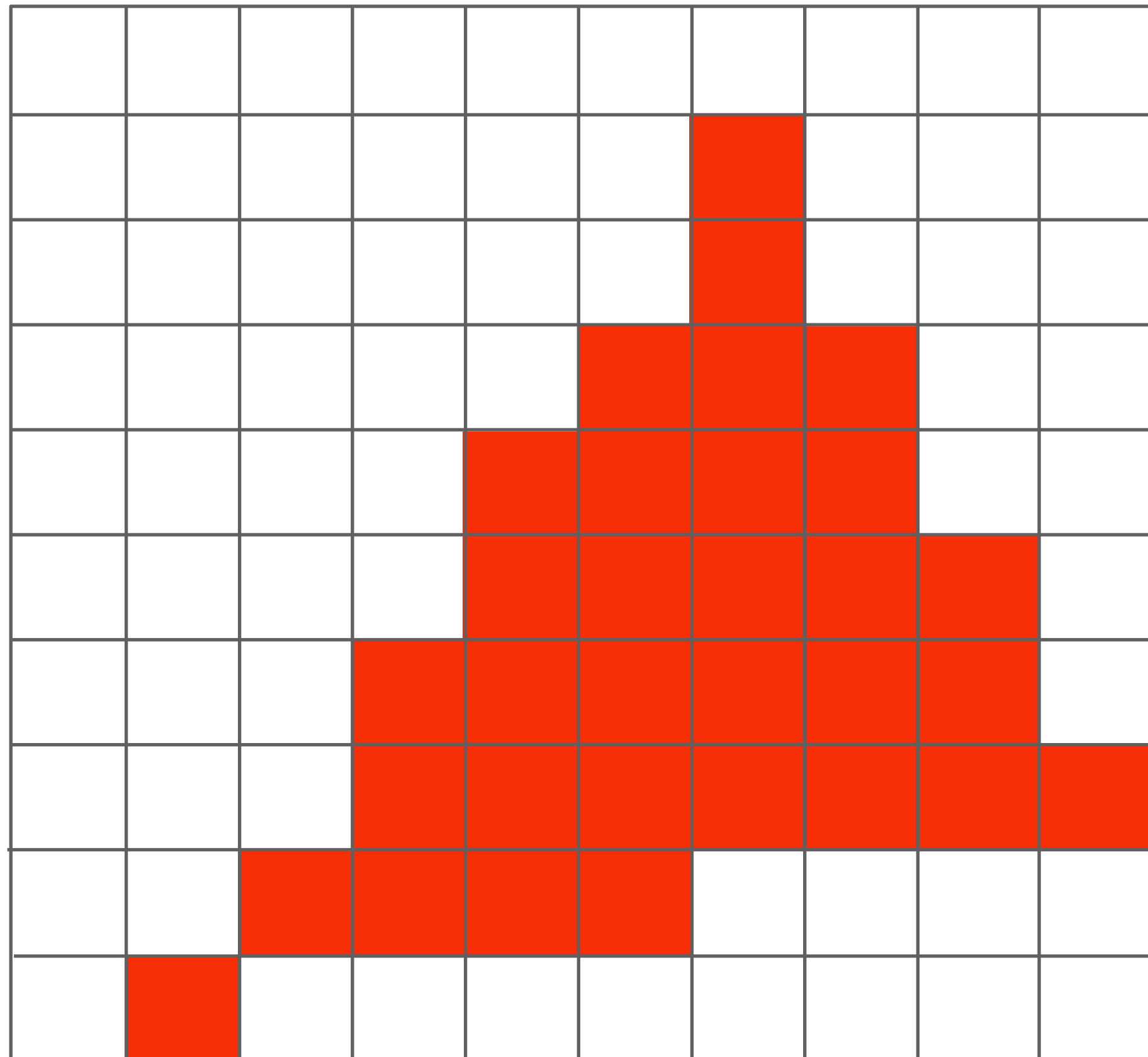


**Given our simplified “square pixel” display assumption, we’ve effectively performed a piecewise constant reconstruction**

# Compare: the continuous triangle function

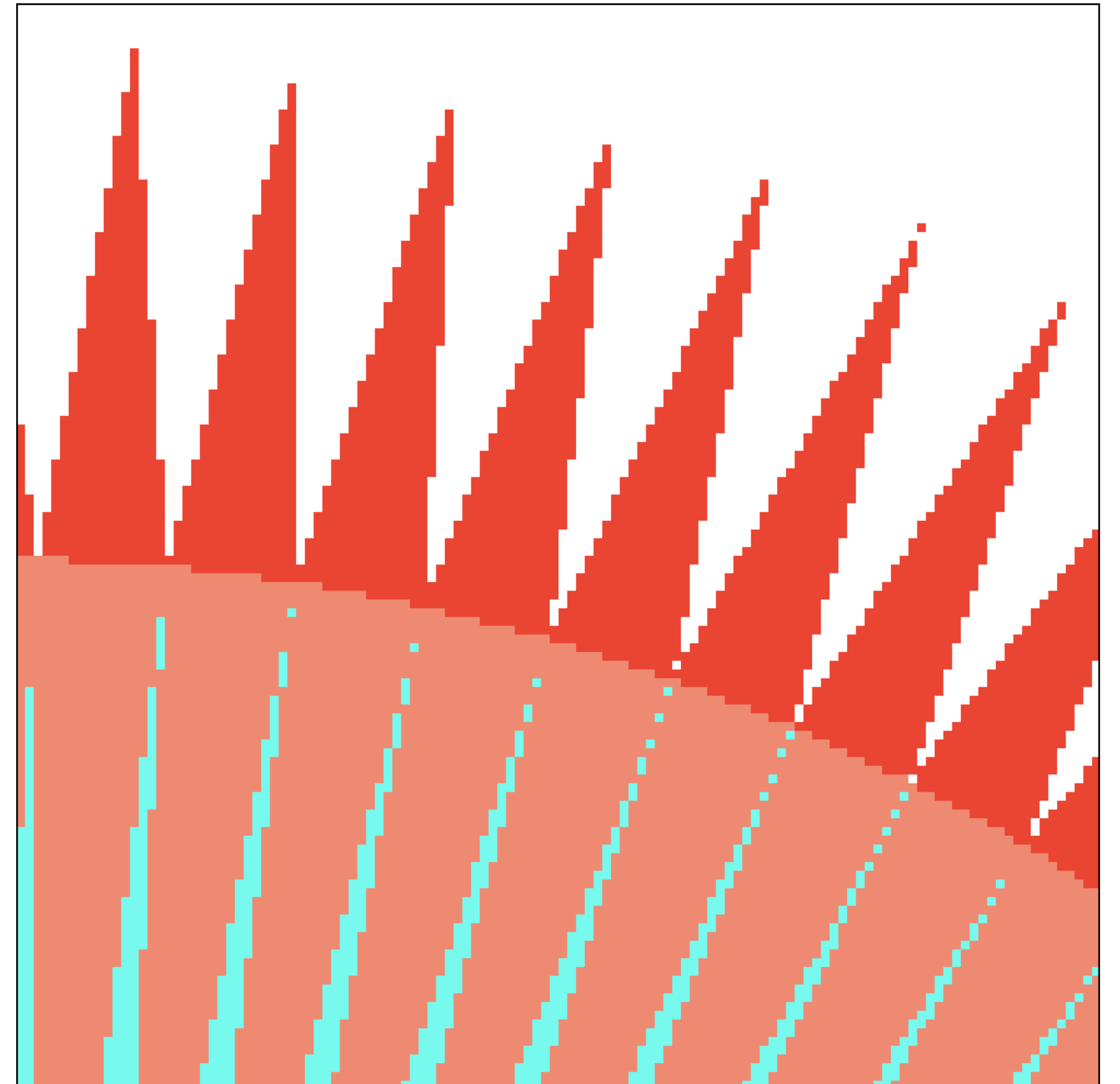
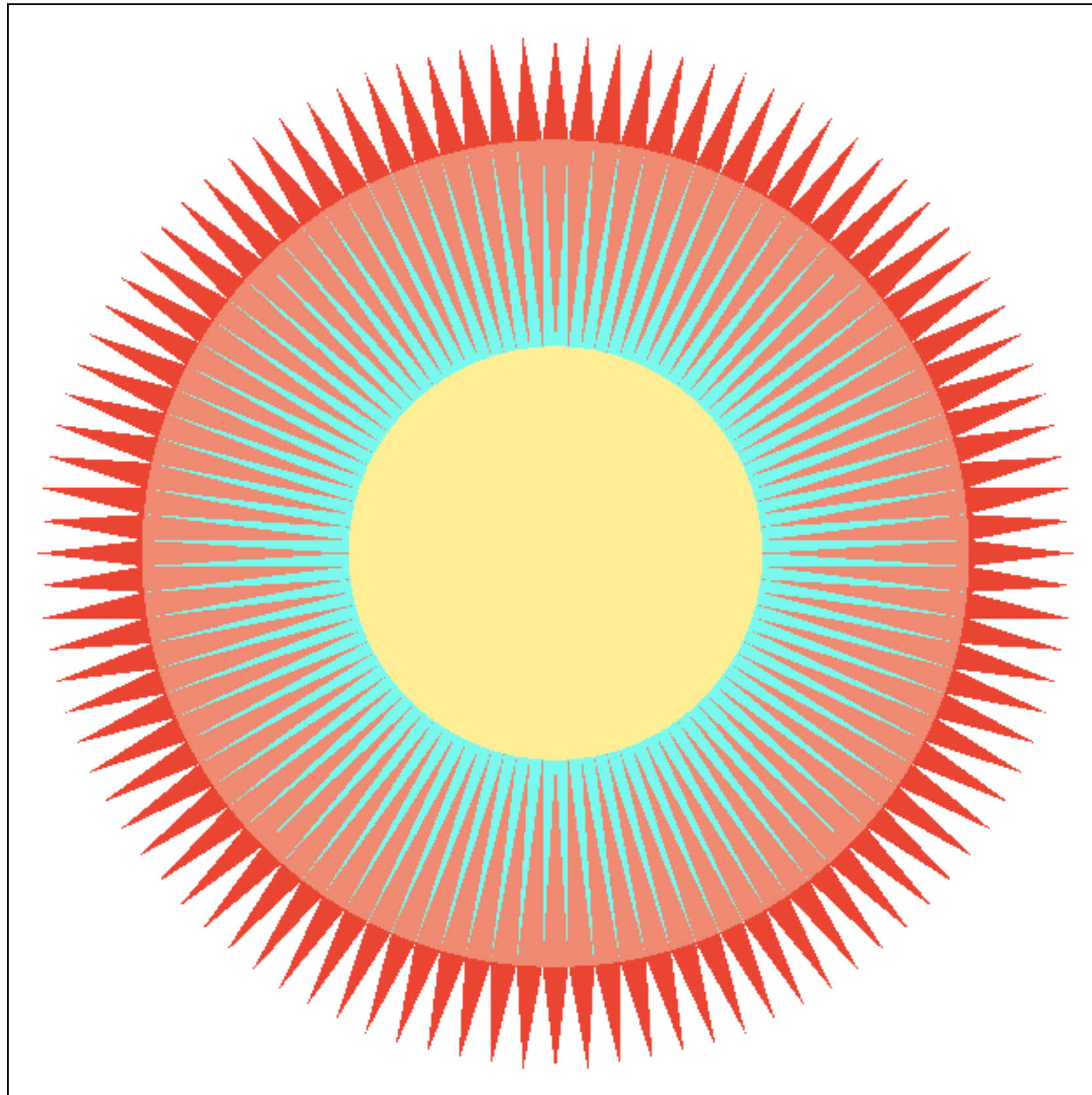


# What's wrong with this picture?



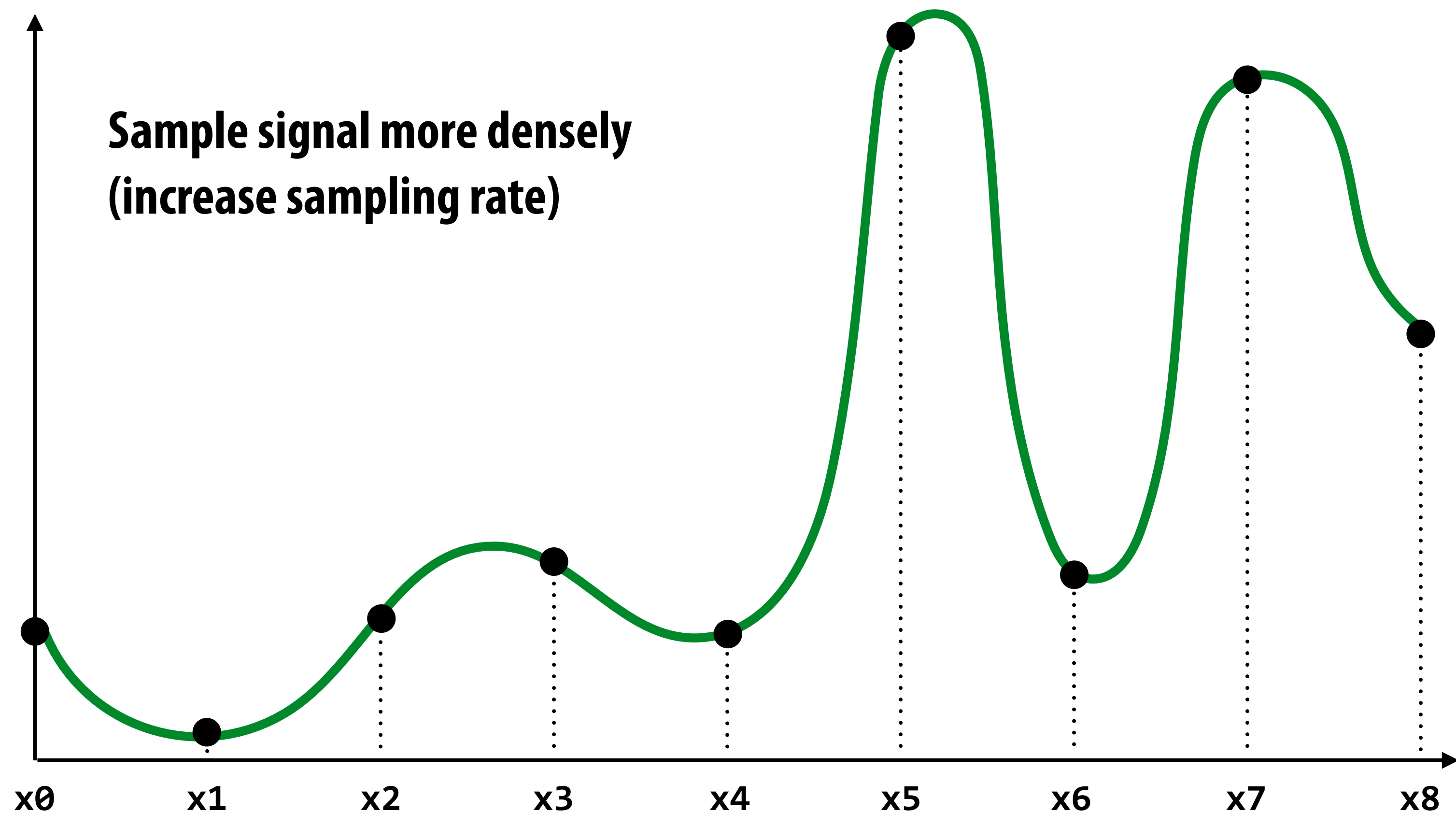
**Jaggies!**

# Jaggies (staircase pattern)

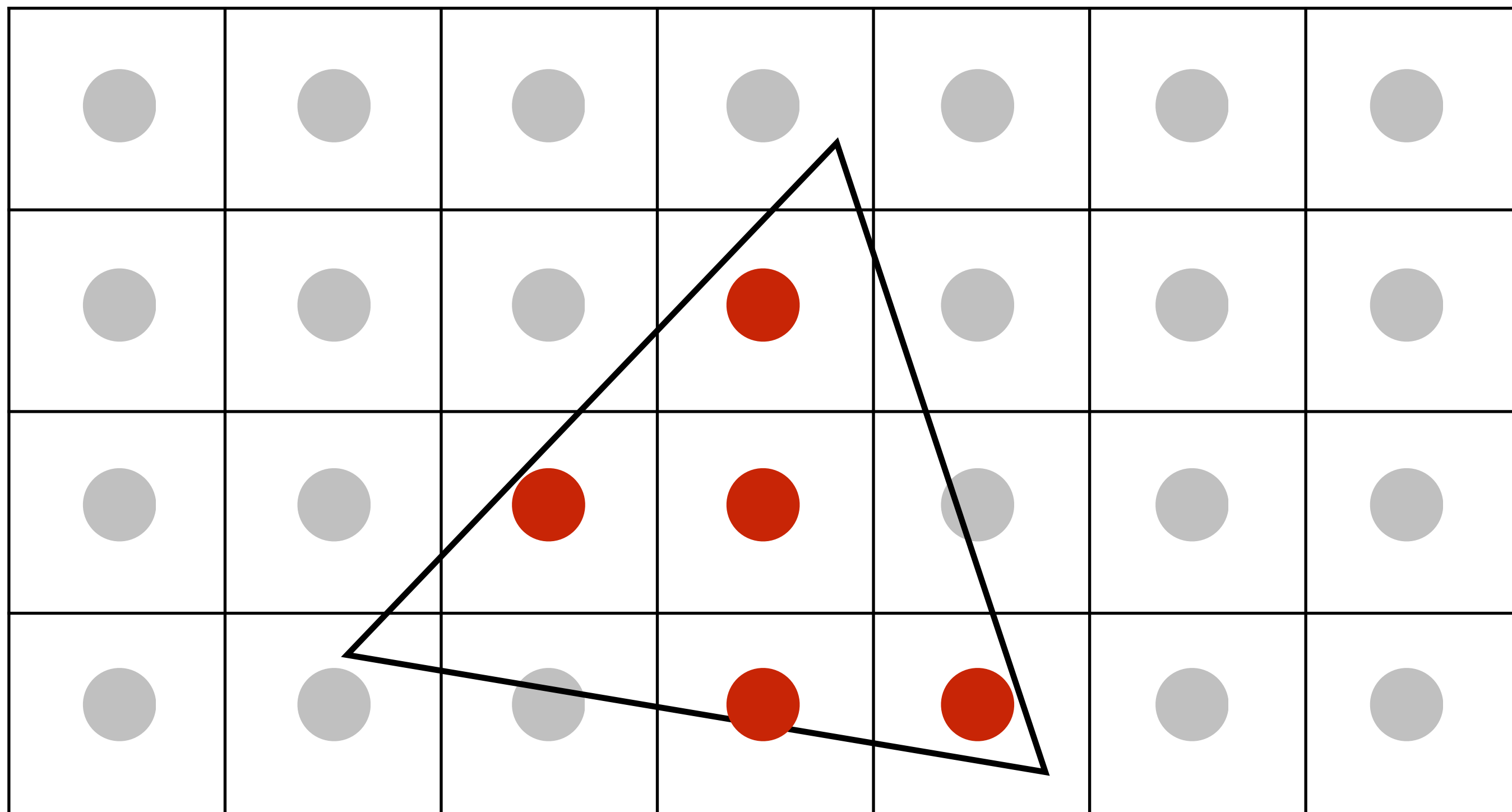


**Is this the best we can do?**

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



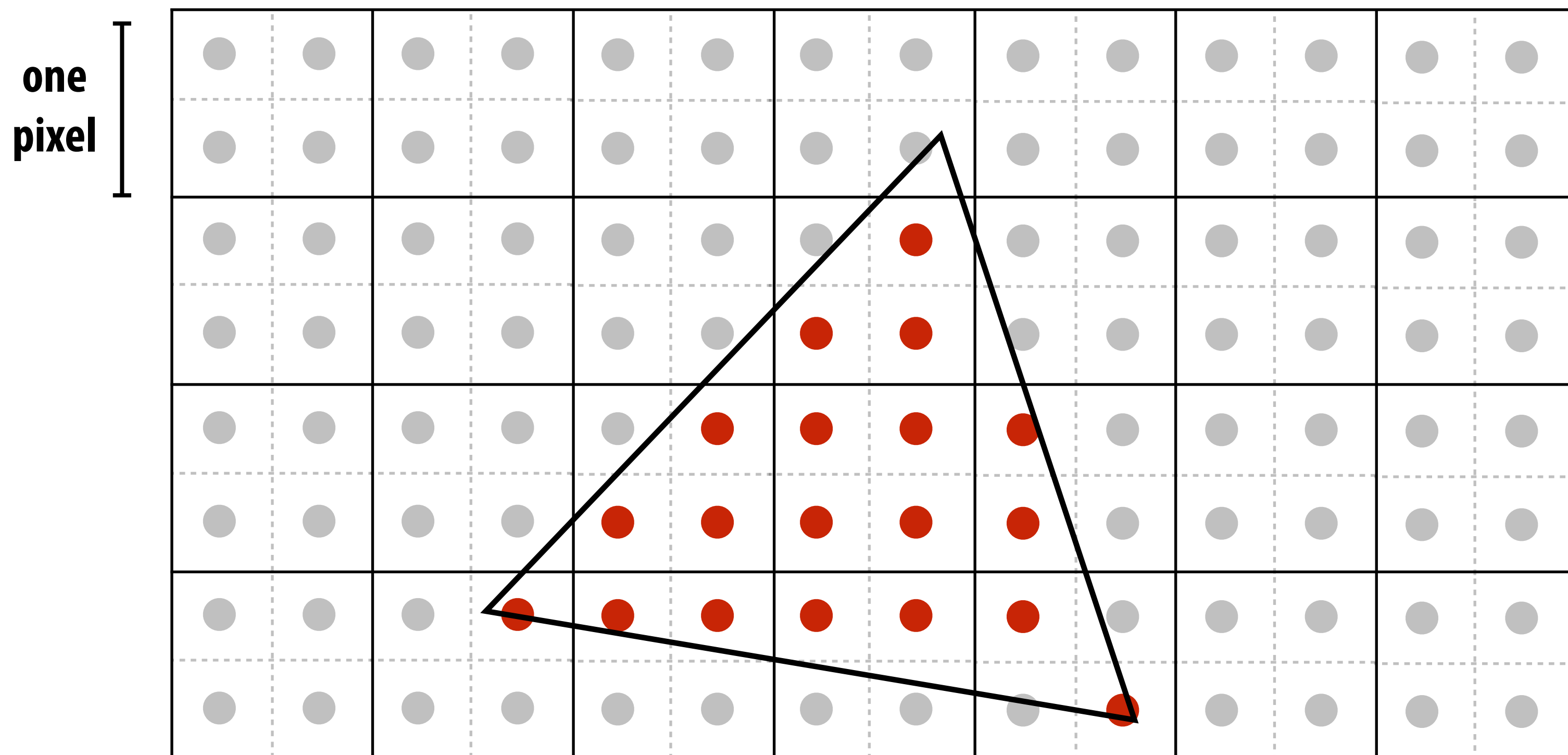
# Point sampling: one sample per pixel



# Supersampling: step 1

Take  $N \times N$  samples in each pixel

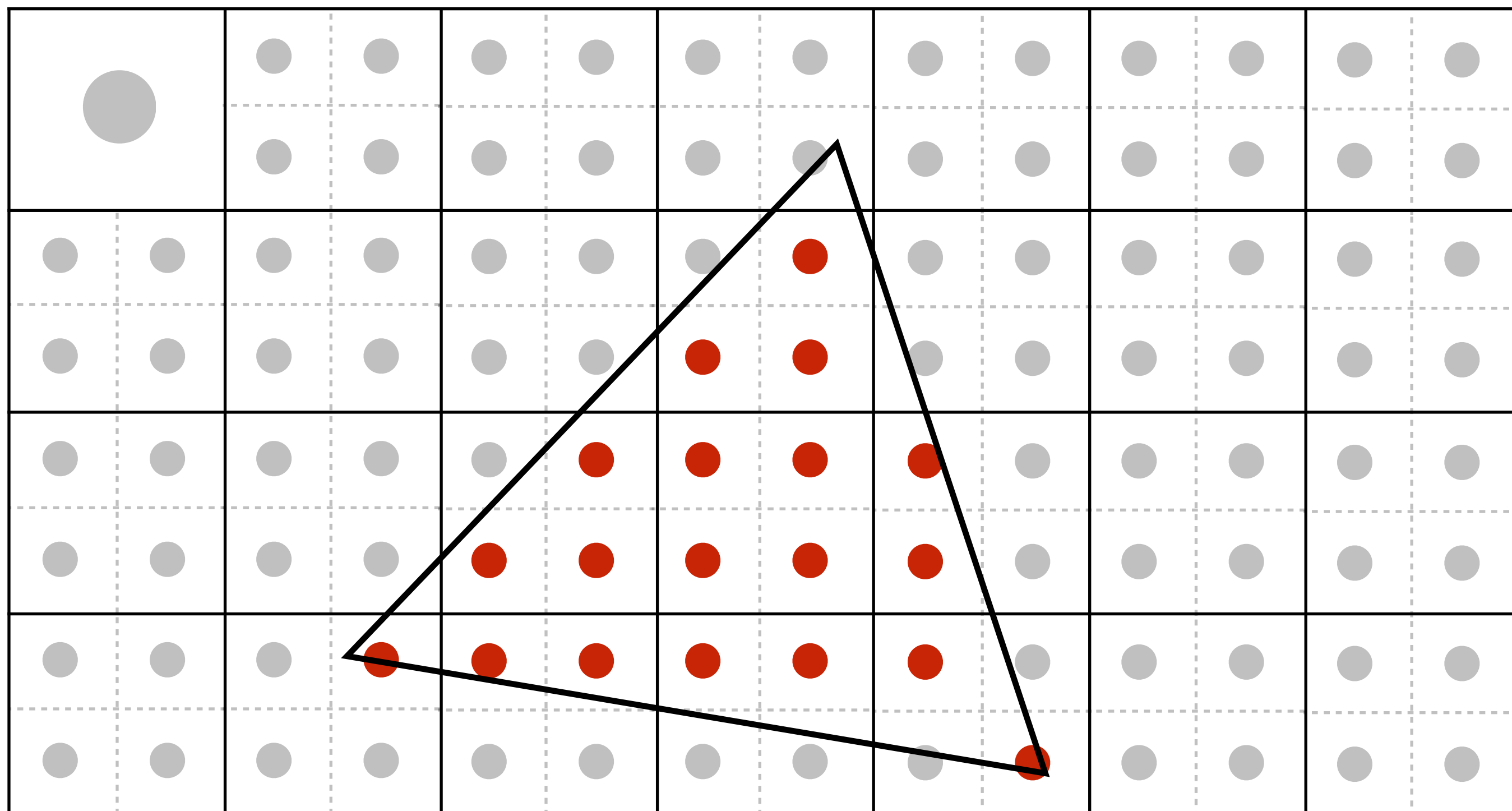
(but... how do we use these samples to drive a display, since there are four times more samples than display pixels!)



**2x2 supersampling**

# Supersampling: step 2

Average the  $N \times N$  samples "inside" each pixel

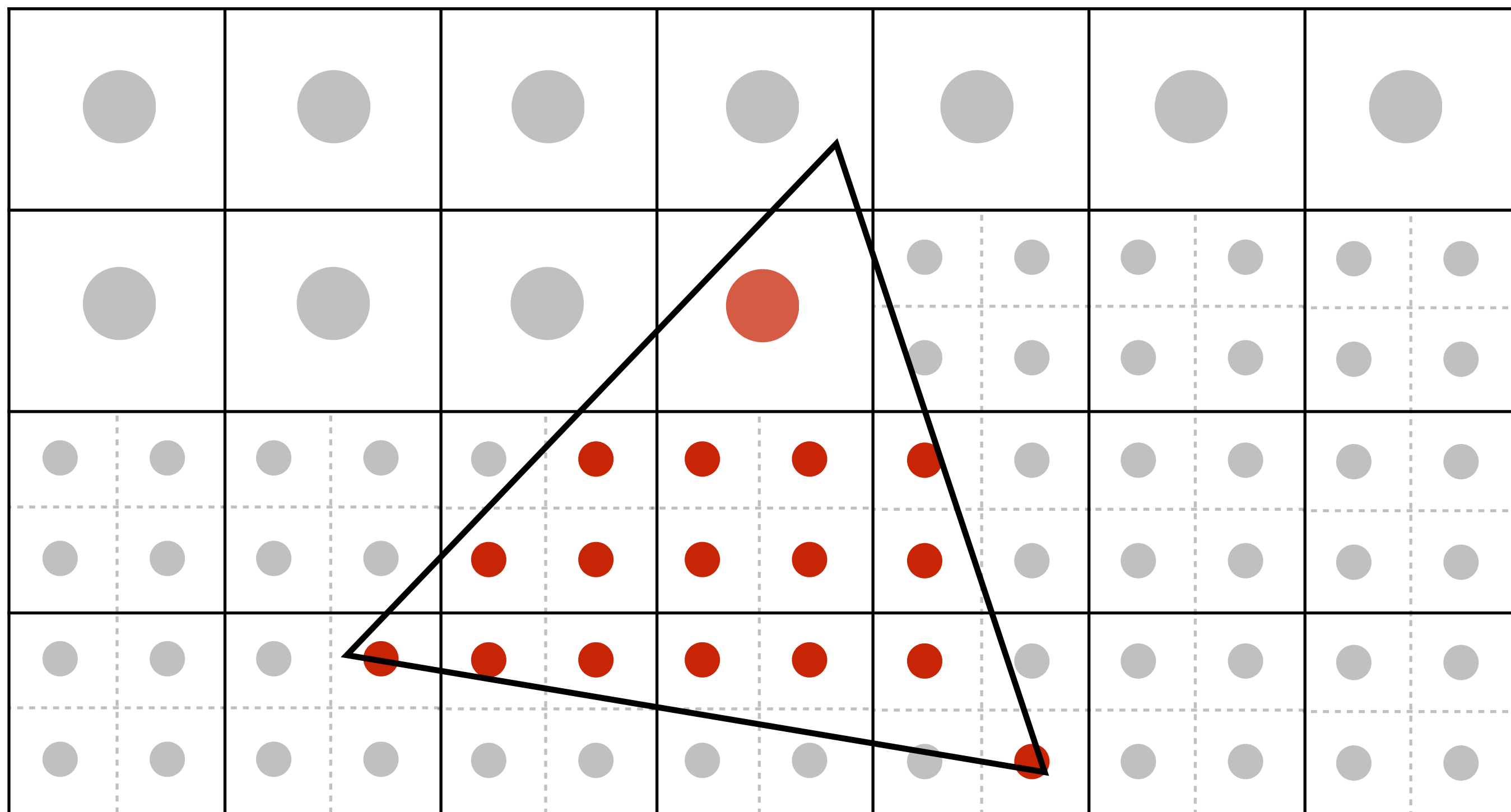


**Averaging down**



# Supersampling: step 2

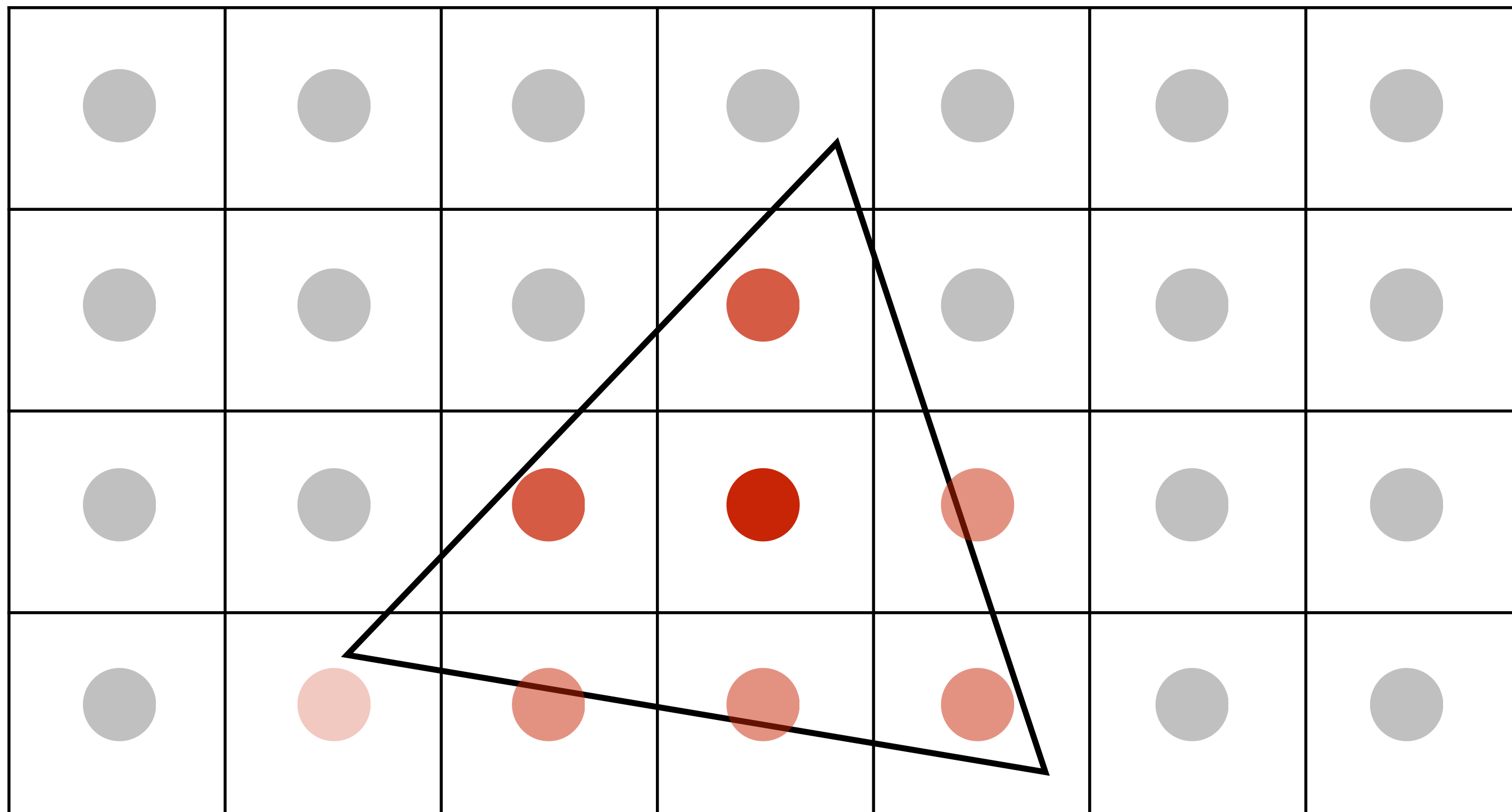
Average the  $N \times N$  samples "inside" each pixel



**Averaging down**

# Supersampling: step 2

Average the  $N \times N$  samples "inside" each pixel

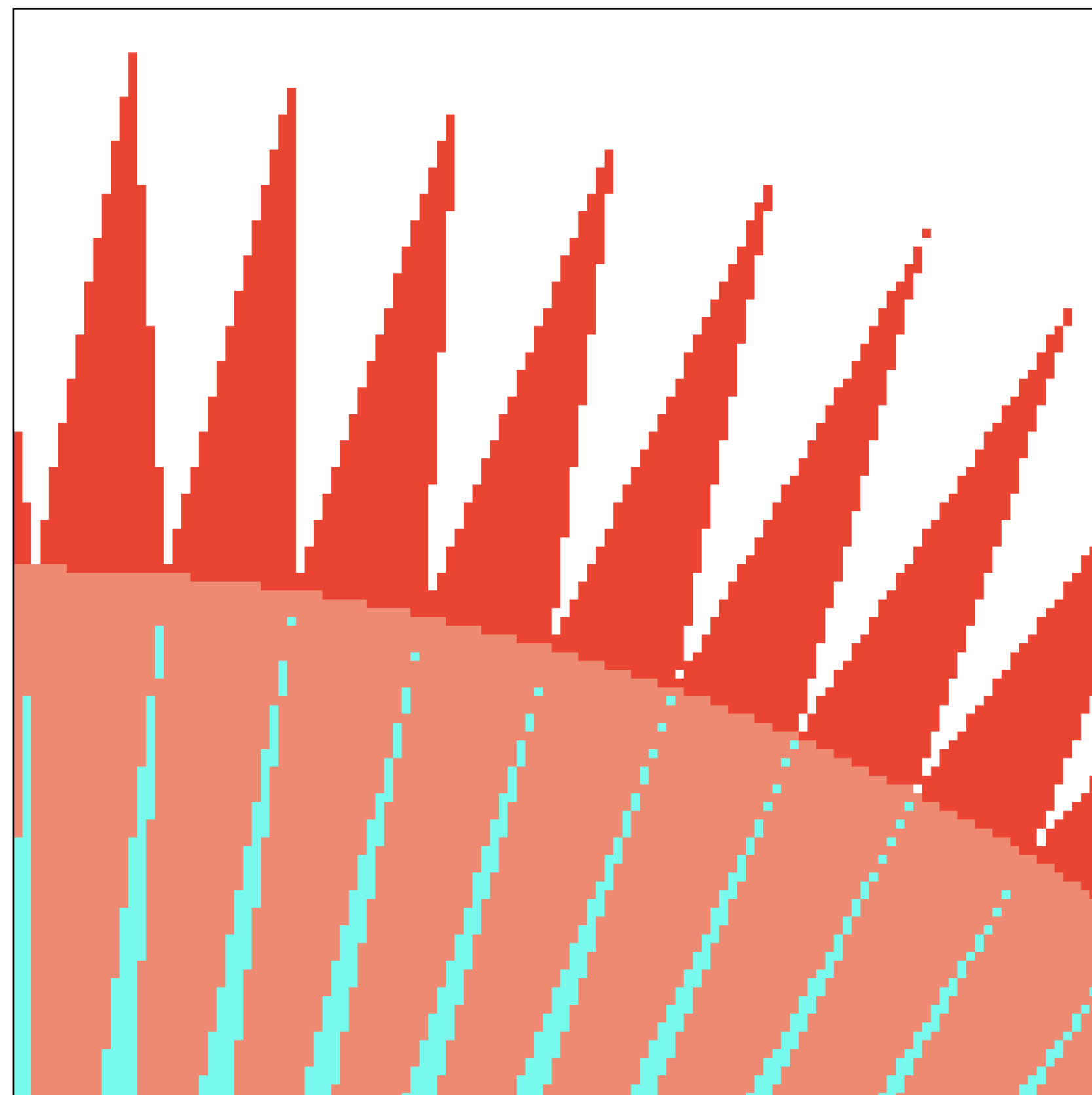
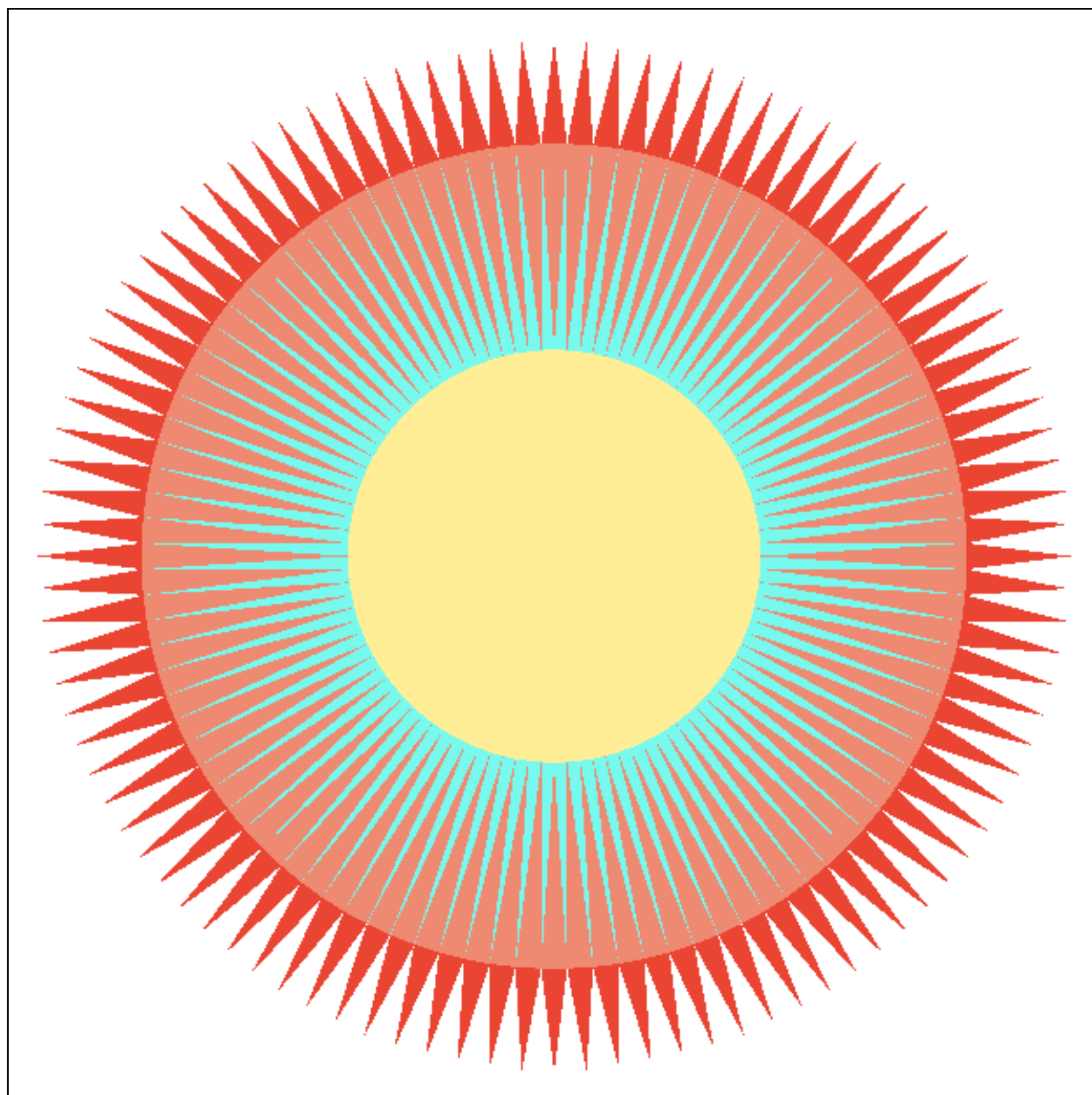


# Supersampling: result

This is the corresponding signal emitted by the display

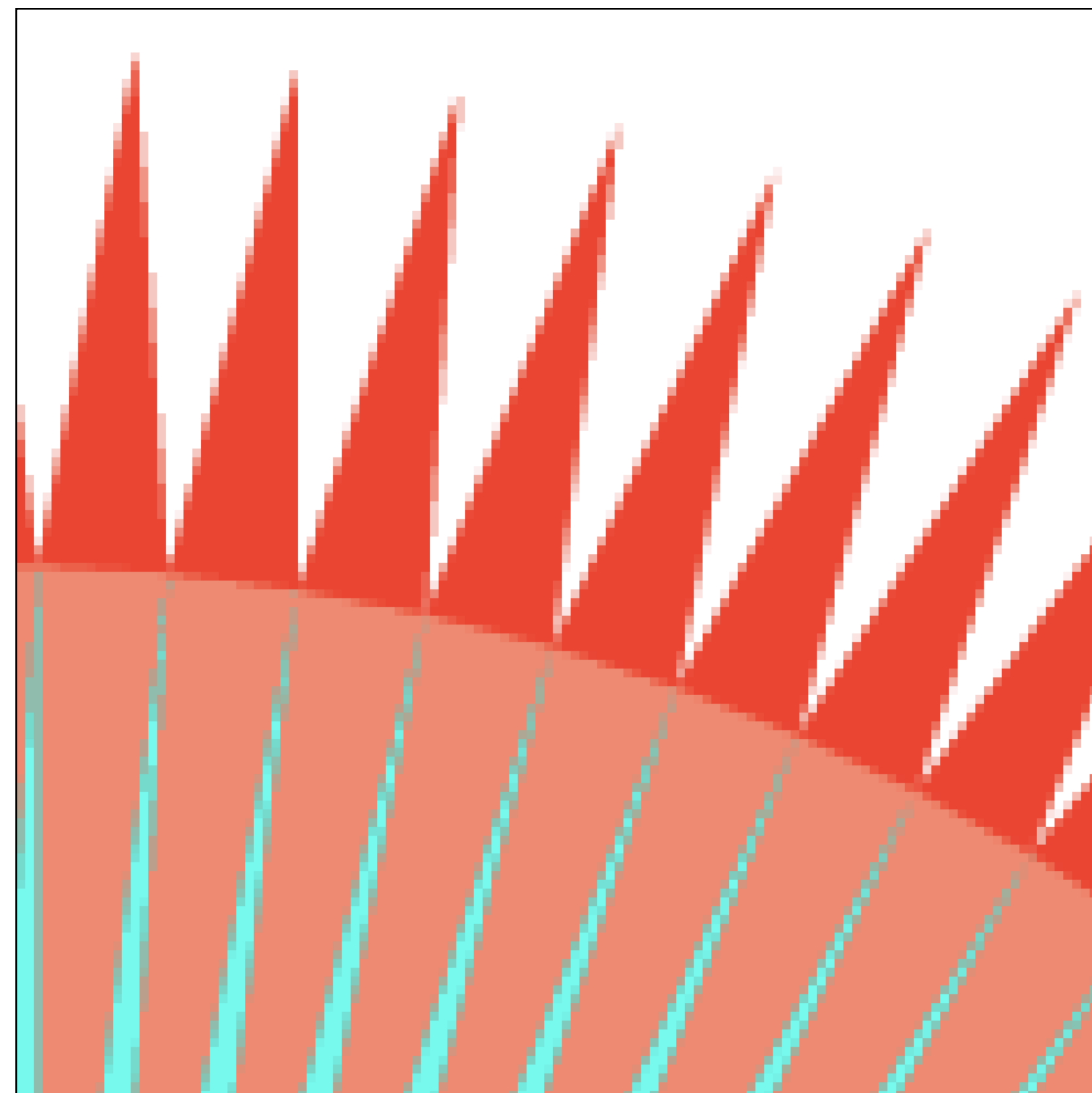
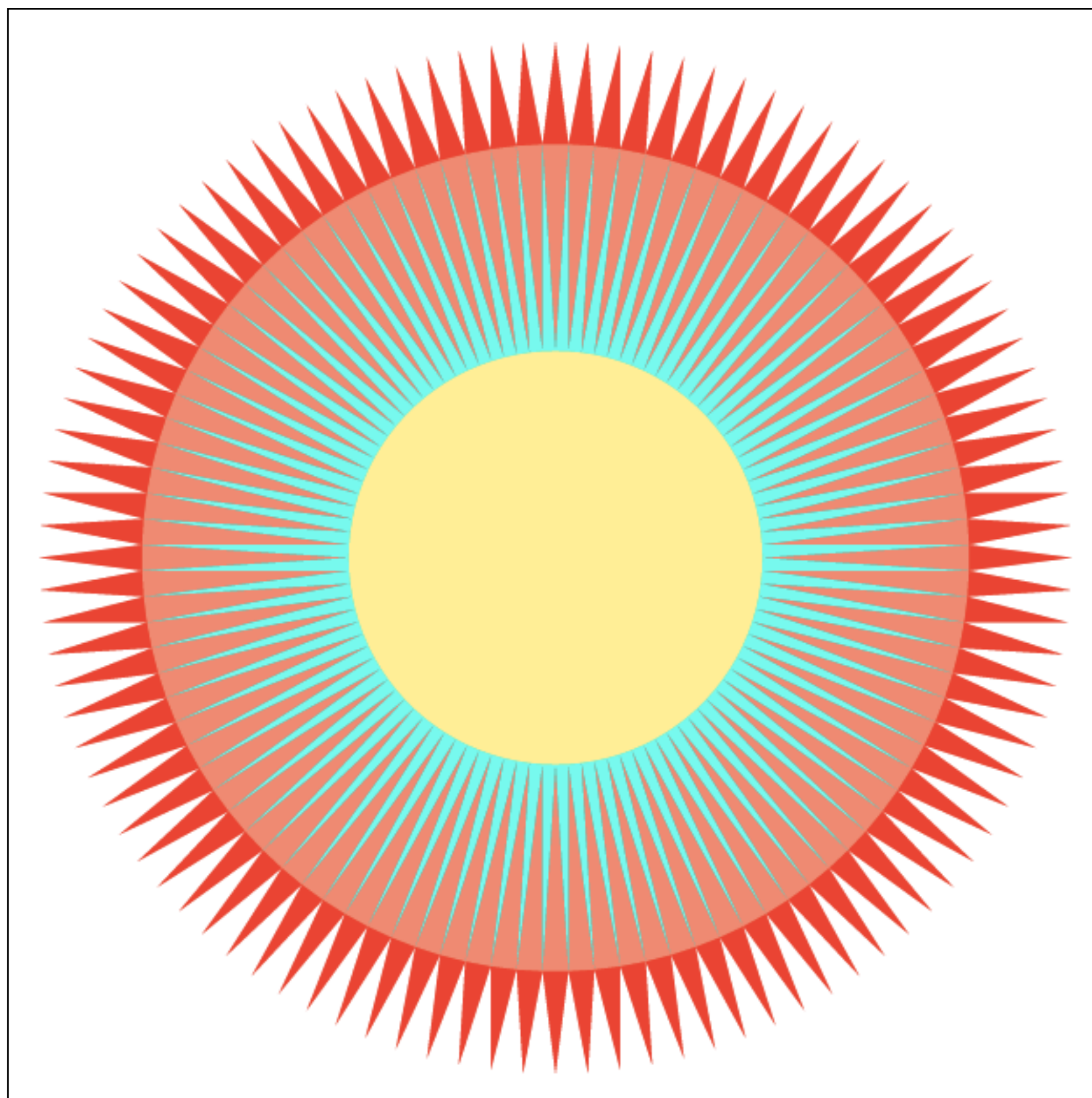
			75%			
		100%	100%	50%		
	25%	50%	50%	50%		

# Point sampling



**One sample per pixel**

# 4x4 supersampling + downsampling

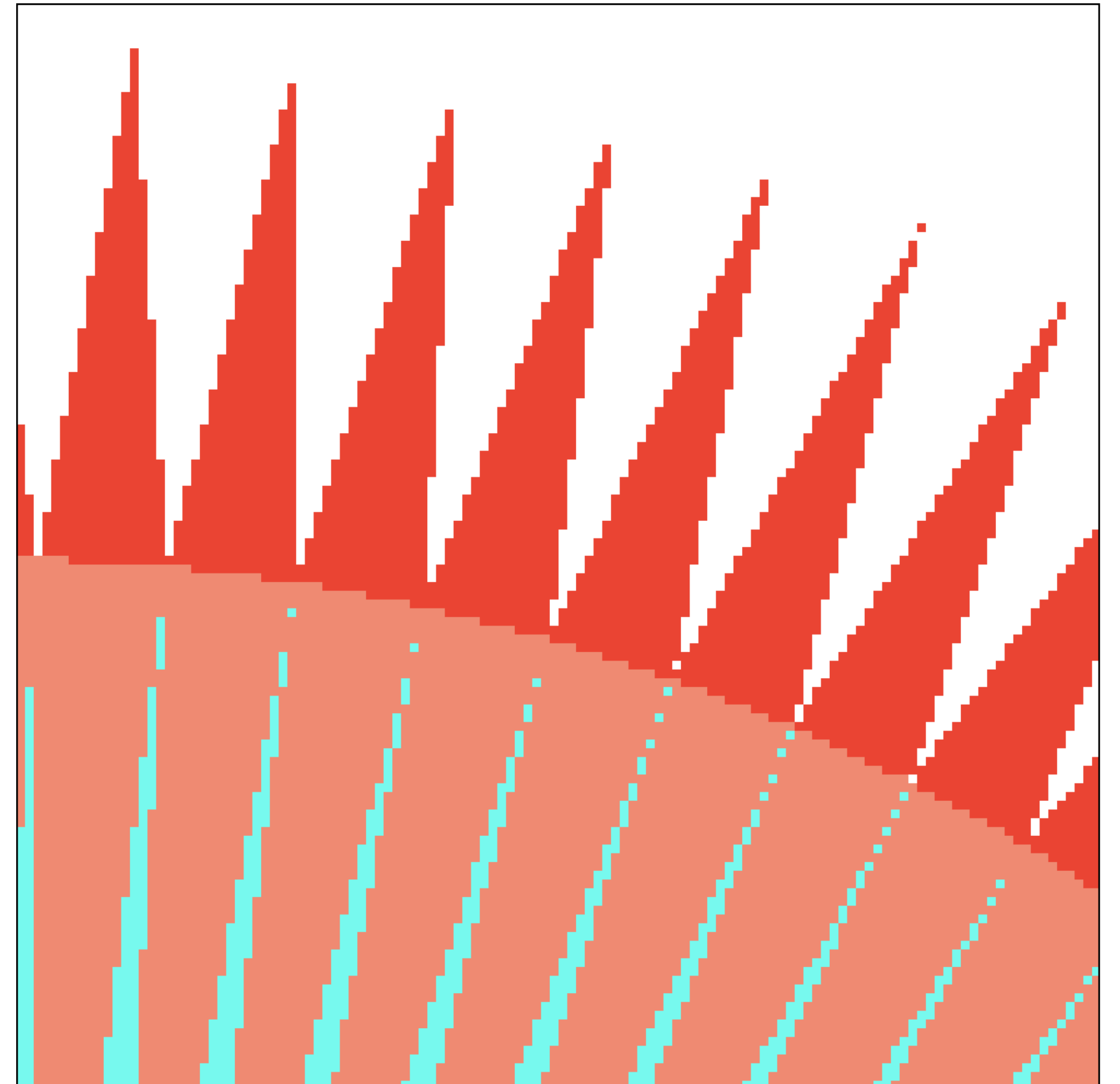
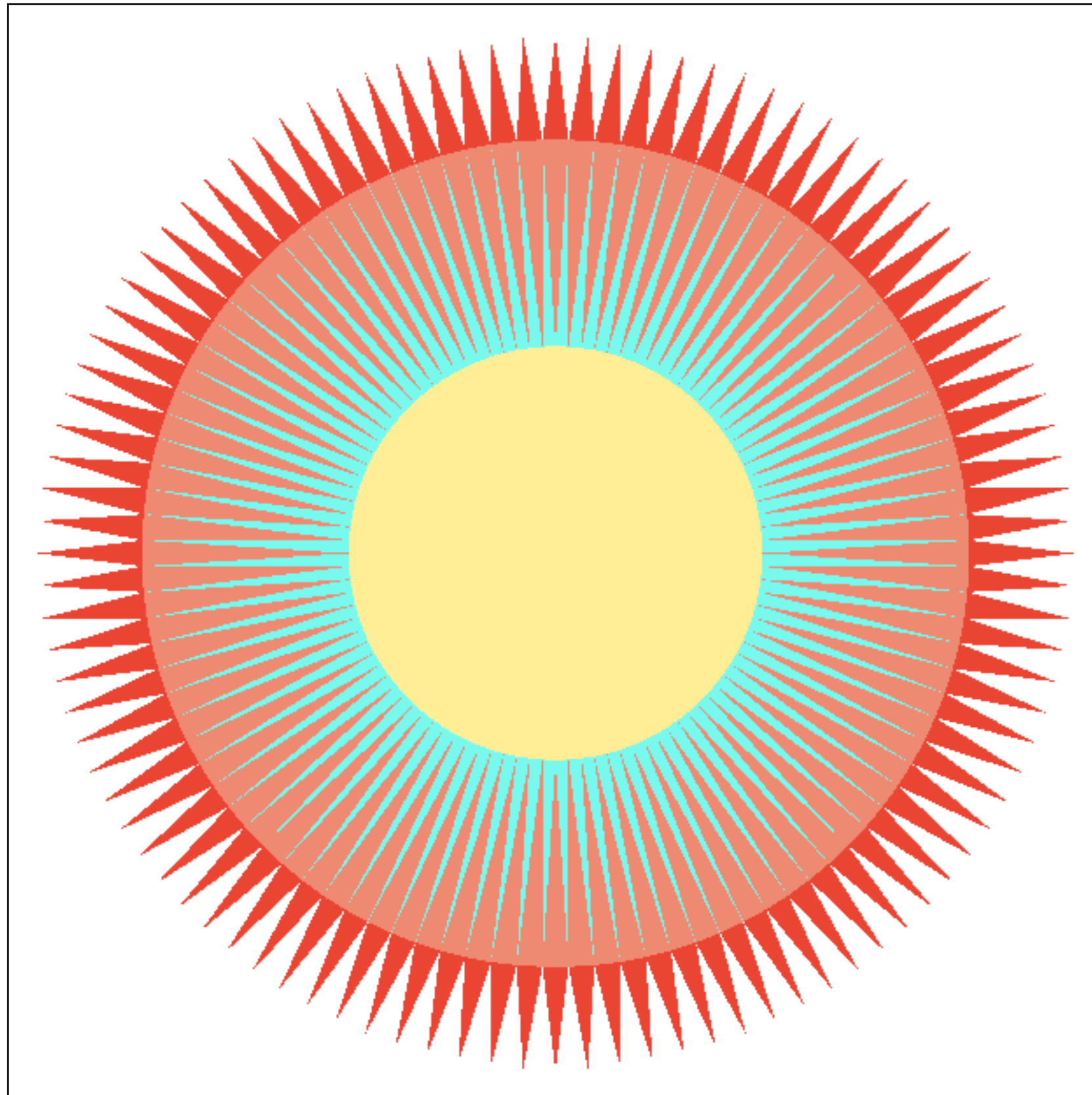


**Pixel value is average of 4x4 samples per pixel**

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

# **More examples of sampling artifacts in computer graphics**

# Jaggies (staircase pattern)



**Is this the best we can do?**



# Moiré patterns in imaging



**Read every sensor pixel**



**Skip odd rows and columns**

lystit.com

# Wagon wheel illusion (false motion)



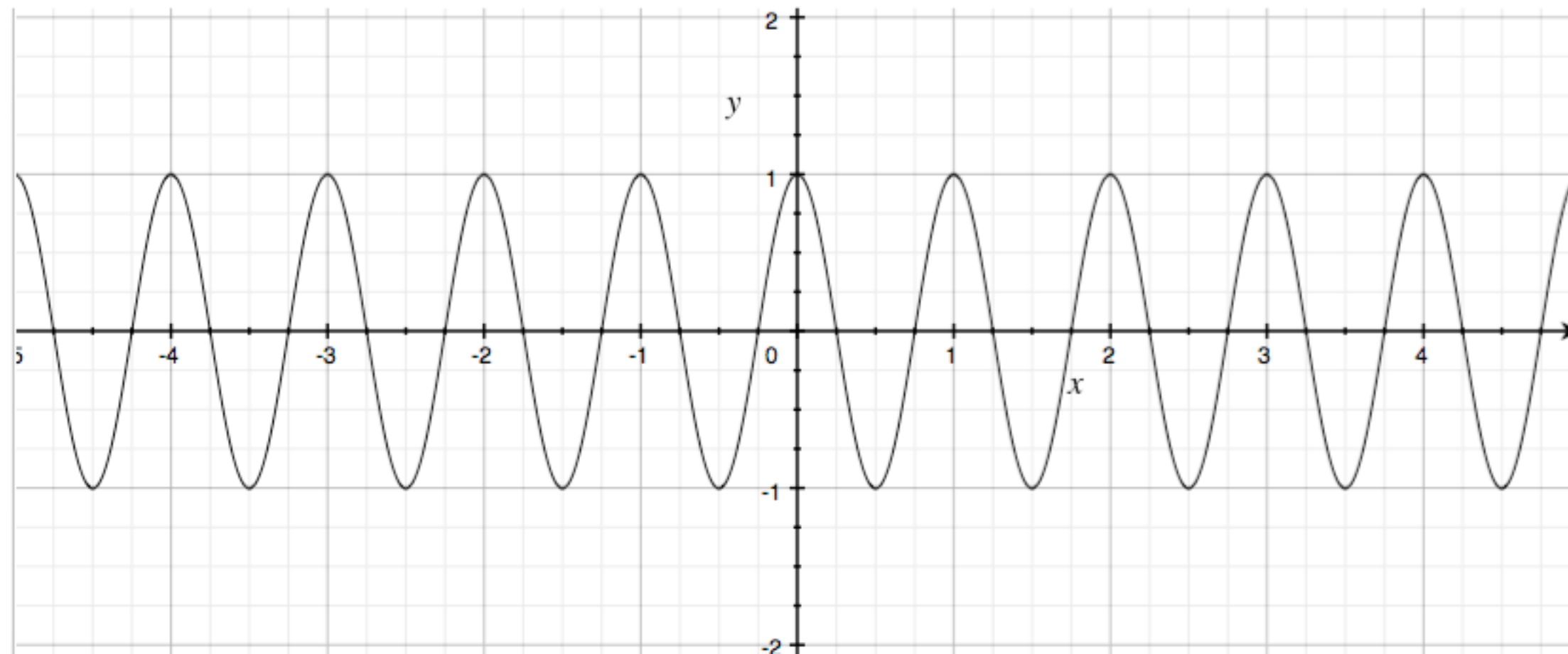
**Camera's frame rate (temporal sampling rate) is too low for rapidly spinning wheel.**

**Created by Jesse Mason, [https://www.youtube.com/watch?v=Q0wzkND\\_ooU](https://www.youtube.com/watch?v=Q0wzkND_ooU)**

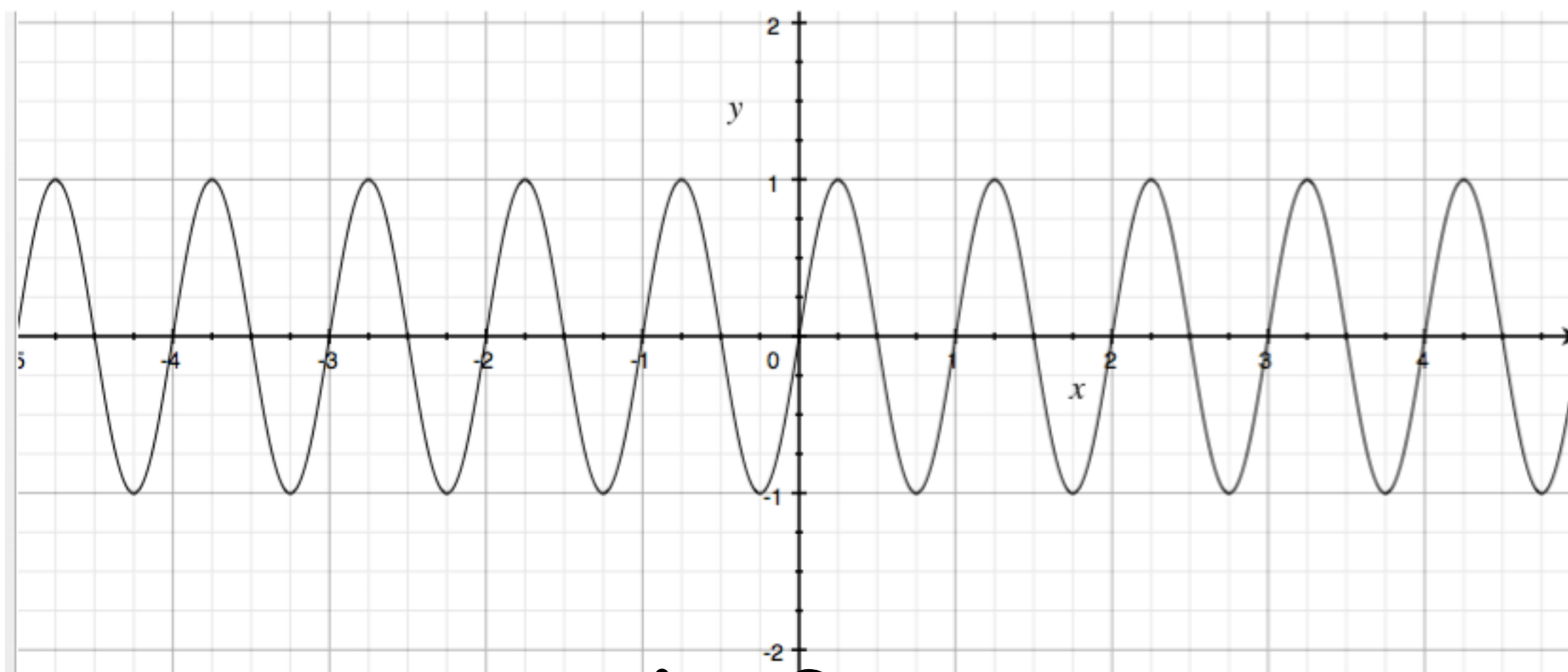
# Sampling artifacts in computer graphics

- **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 too sparsely**

# Sines and cosines



$$\cos 2\pi x$$

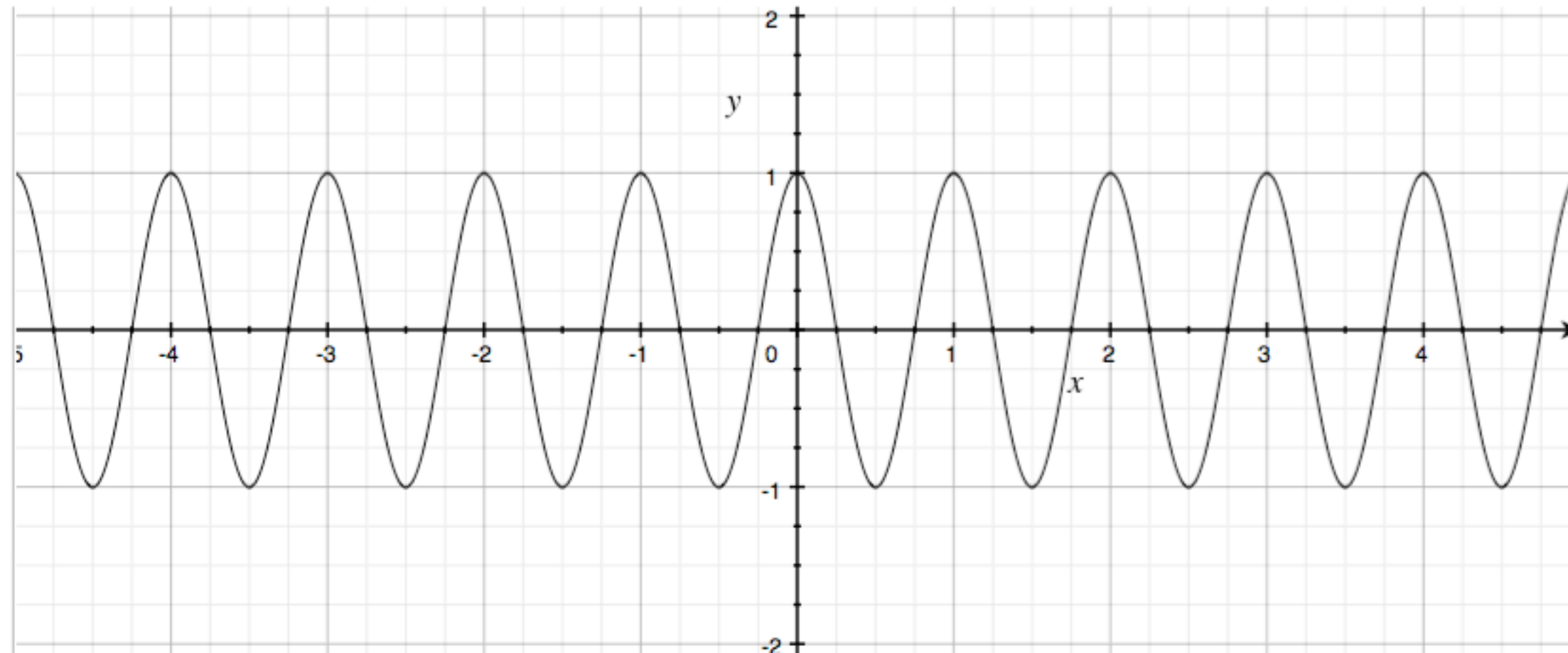


$$\sin 2\pi x$$

# Frequencies

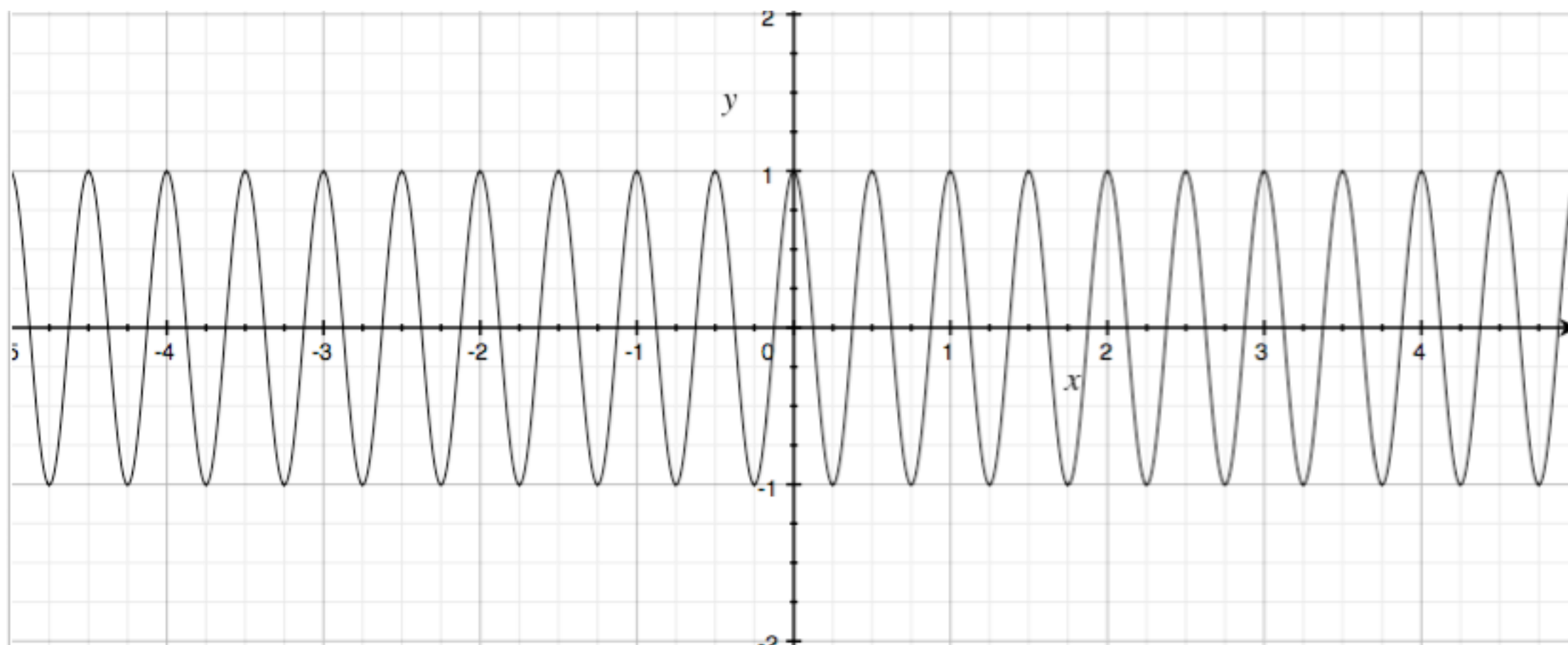
$$\cos 2\pi f x$$

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



$$f = 1$$

$$\cos 2\pi x$$

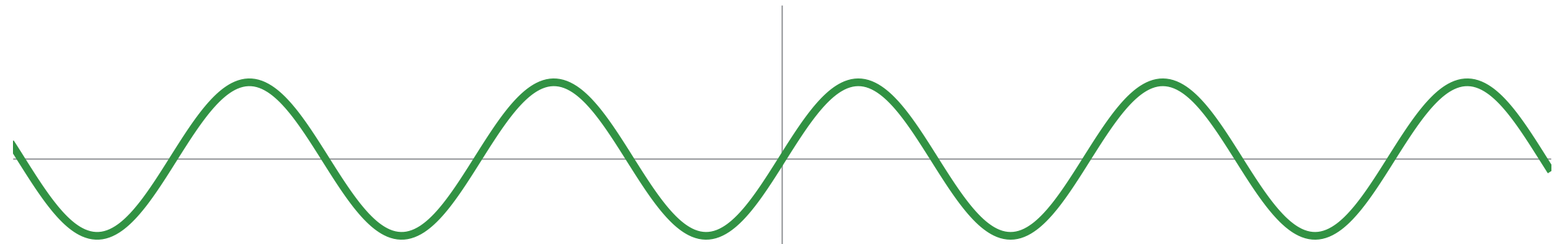


$$f = 2$$

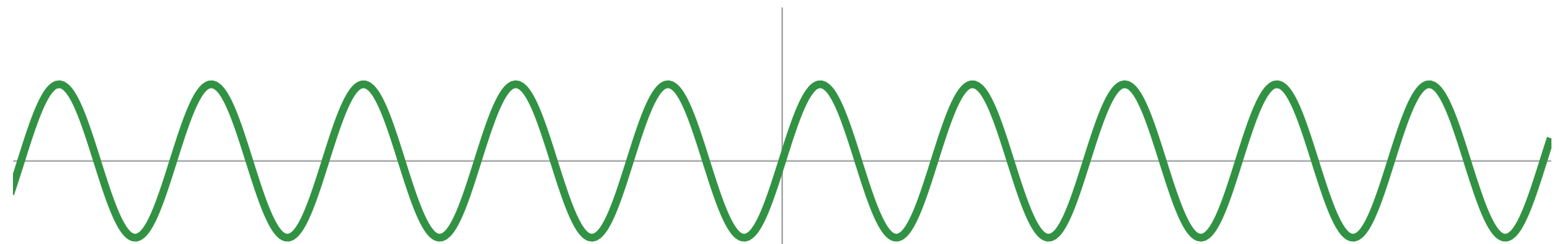
$$\cos 4\pi x$$

# Representing sound as a superposition of frequencies

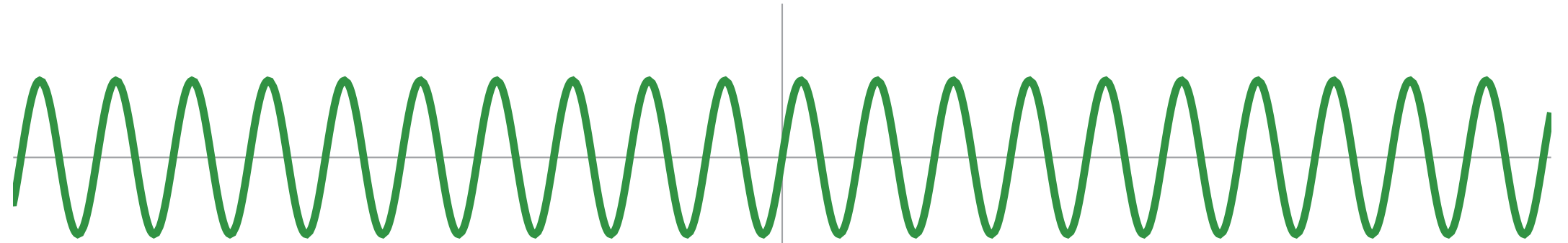
$$f_1(x) = \sin(\pi x)$$



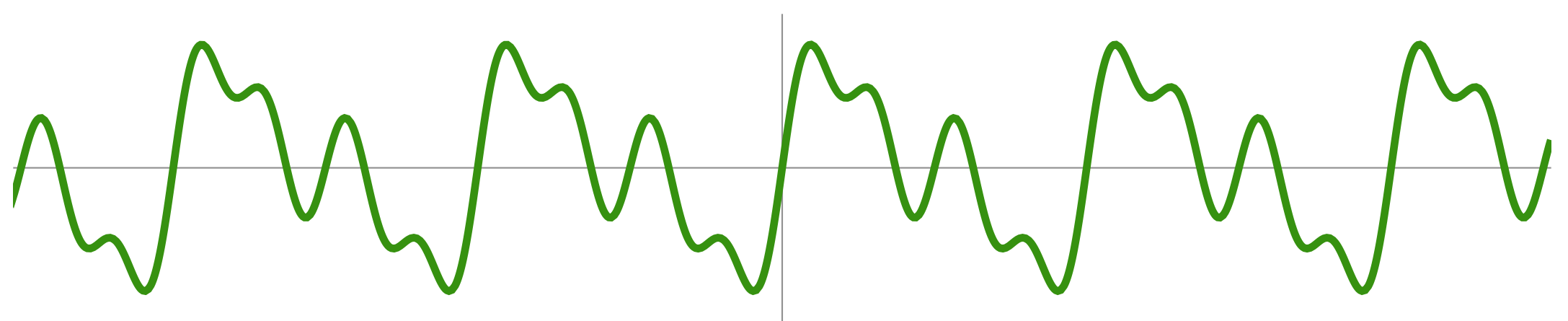
$$f_2(x) = \sin(2\pi x)$$



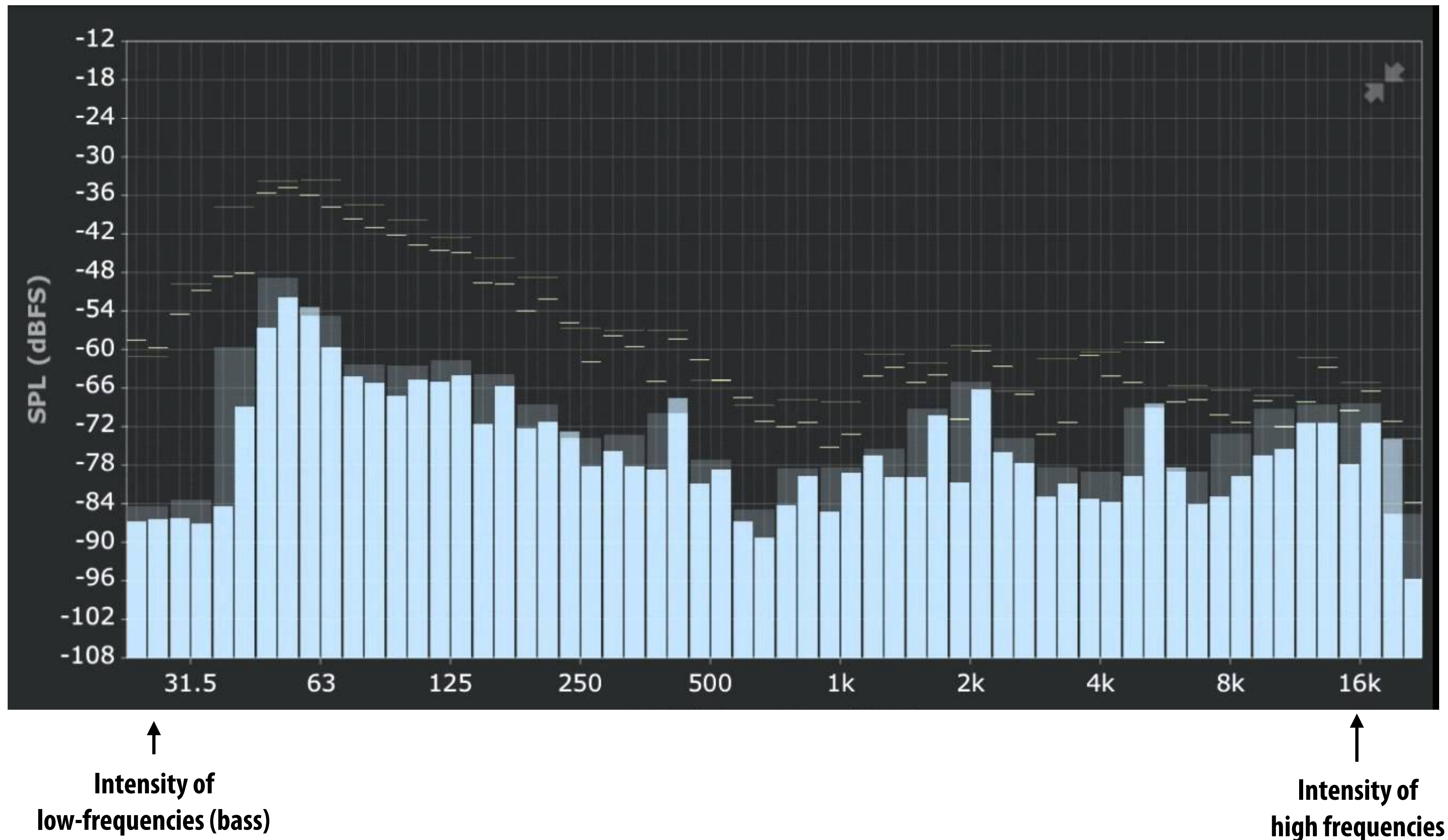
$$f_4(x) = \sin(4\pi x)$$



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



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



# **How to compute frequency-domain representation of a signal?**

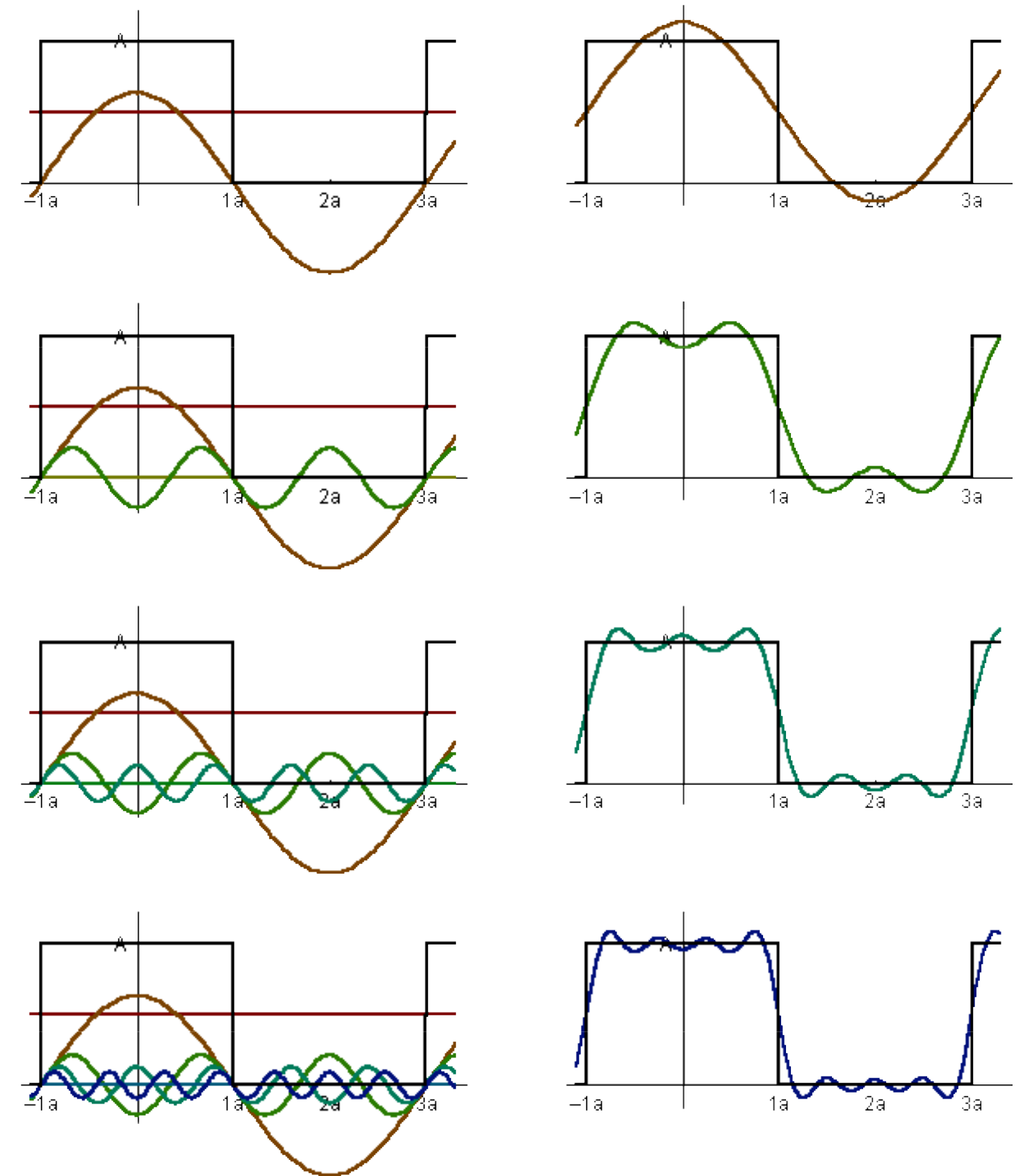


# Fourier transform

Represent a function as a weighted sum of sines and cosines



Joseph Fourier 1768 - 1830



$$f(x) = \frac{A}{2} + \frac{2A \cos(t\omega)}{\pi} - \frac{2A \cos(3t\omega)}{3\pi} + \frac{2A \cos(5t\omega)}{5\pi} - \frac{2A \cos(7t\omega)}{7\pi} + \dots$$

# Fourier transform

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

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

**Recall:**

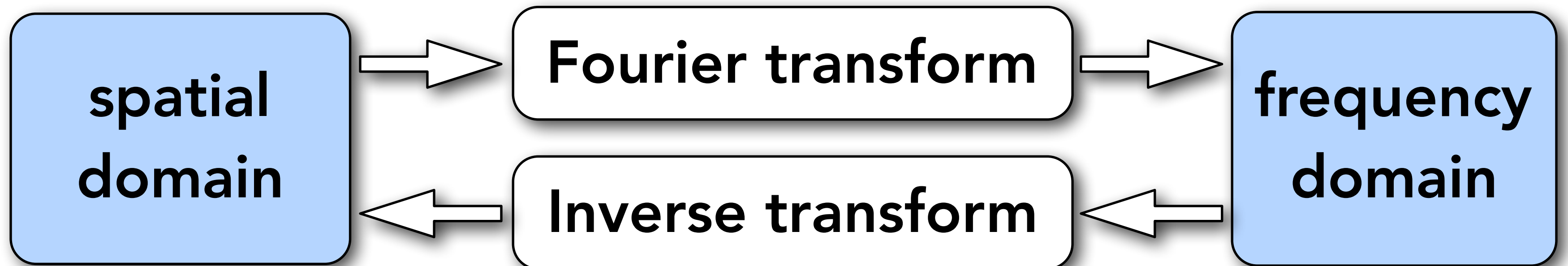
$$e^{ix} = \cos x + i \sin x$$

- **2D form:**

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

# Fourier transform decomposes a signal into its constituent frequencies

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

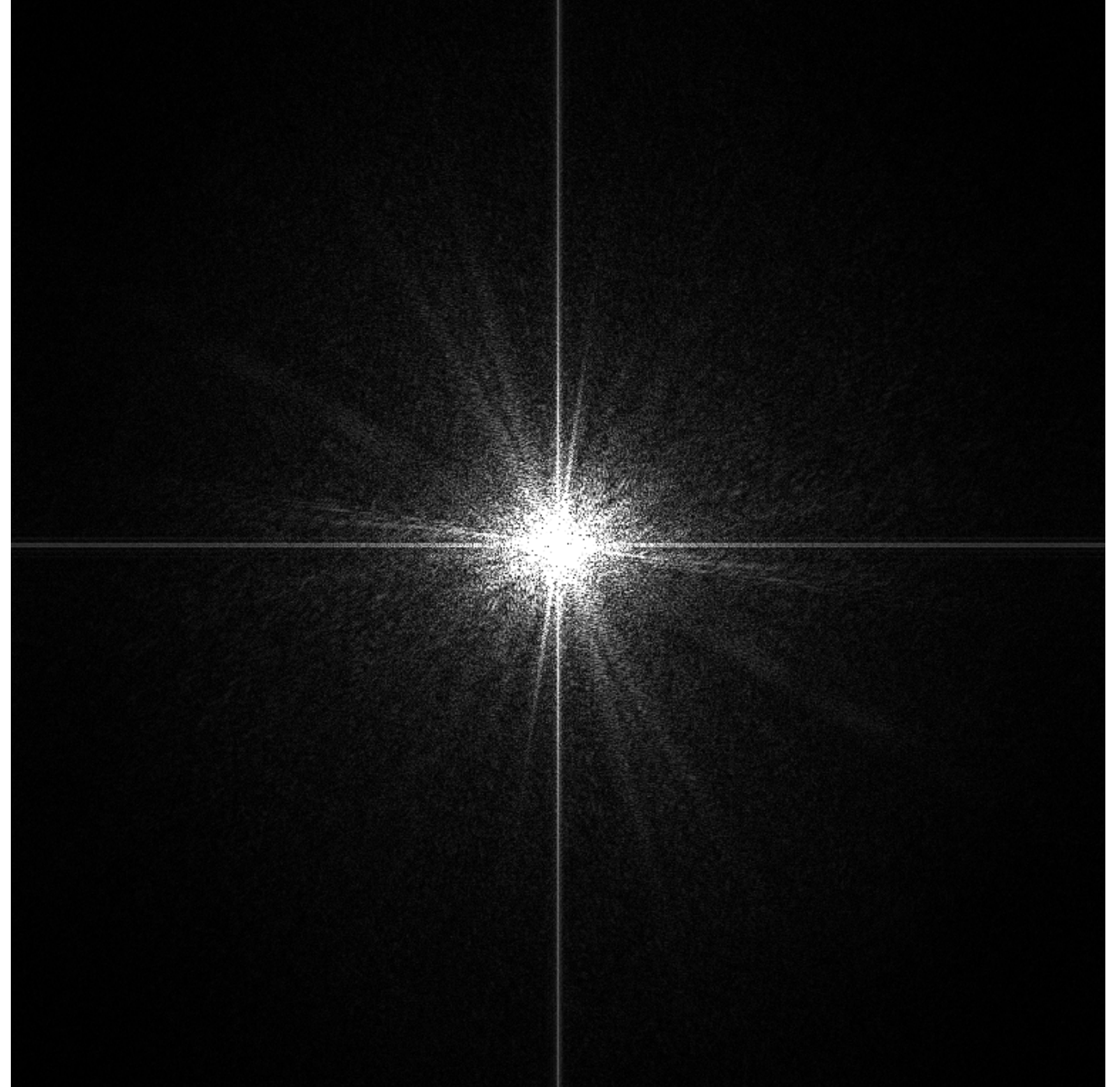


$$f(x) = \int_{-\infty}^{\infty} F(\omega) e^{2\pi i \omega x} d\omega$$

# Visualizing the frequency content of images



**Spatial domain result**

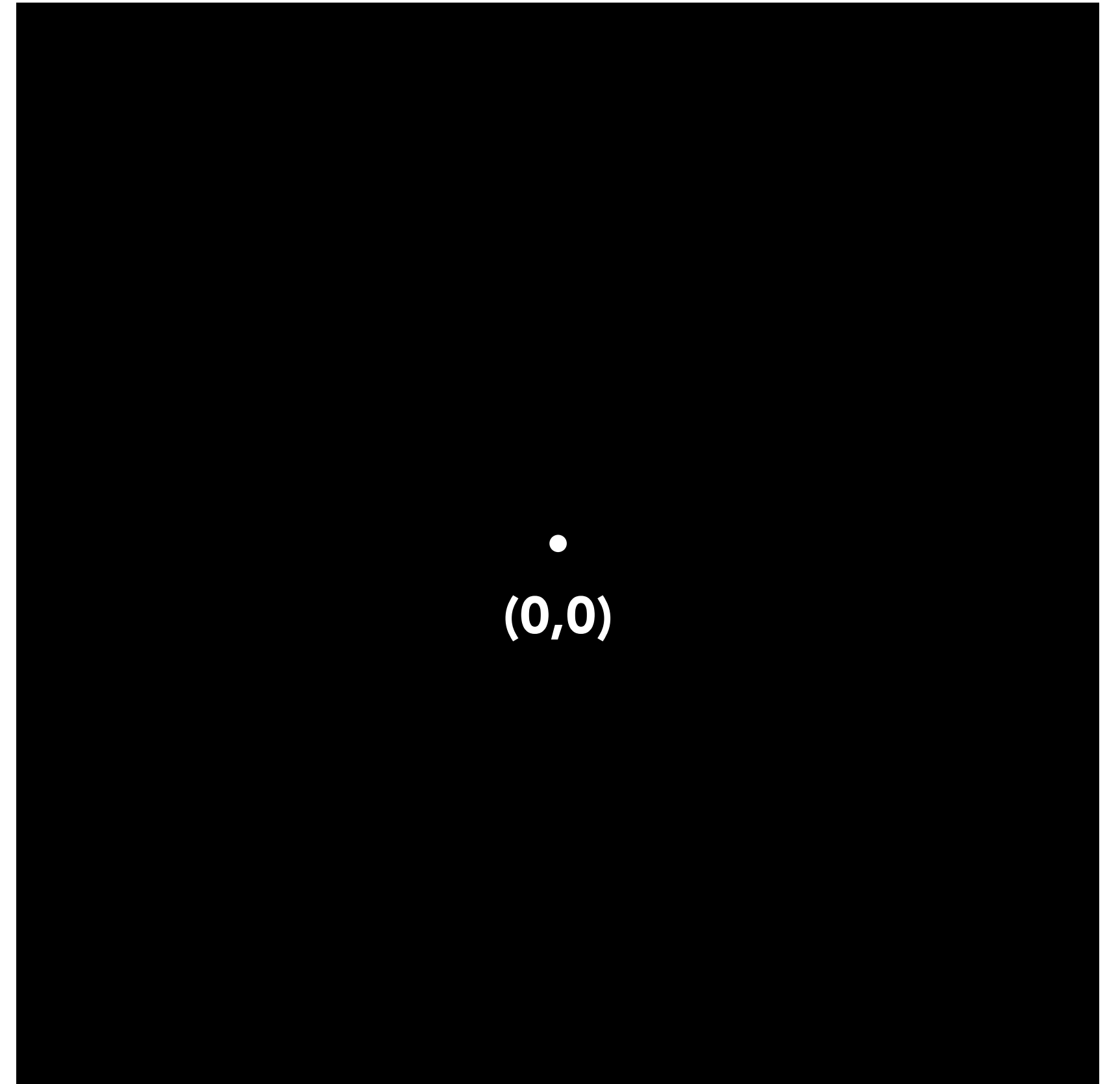


**Spectrum**

# Constant signal

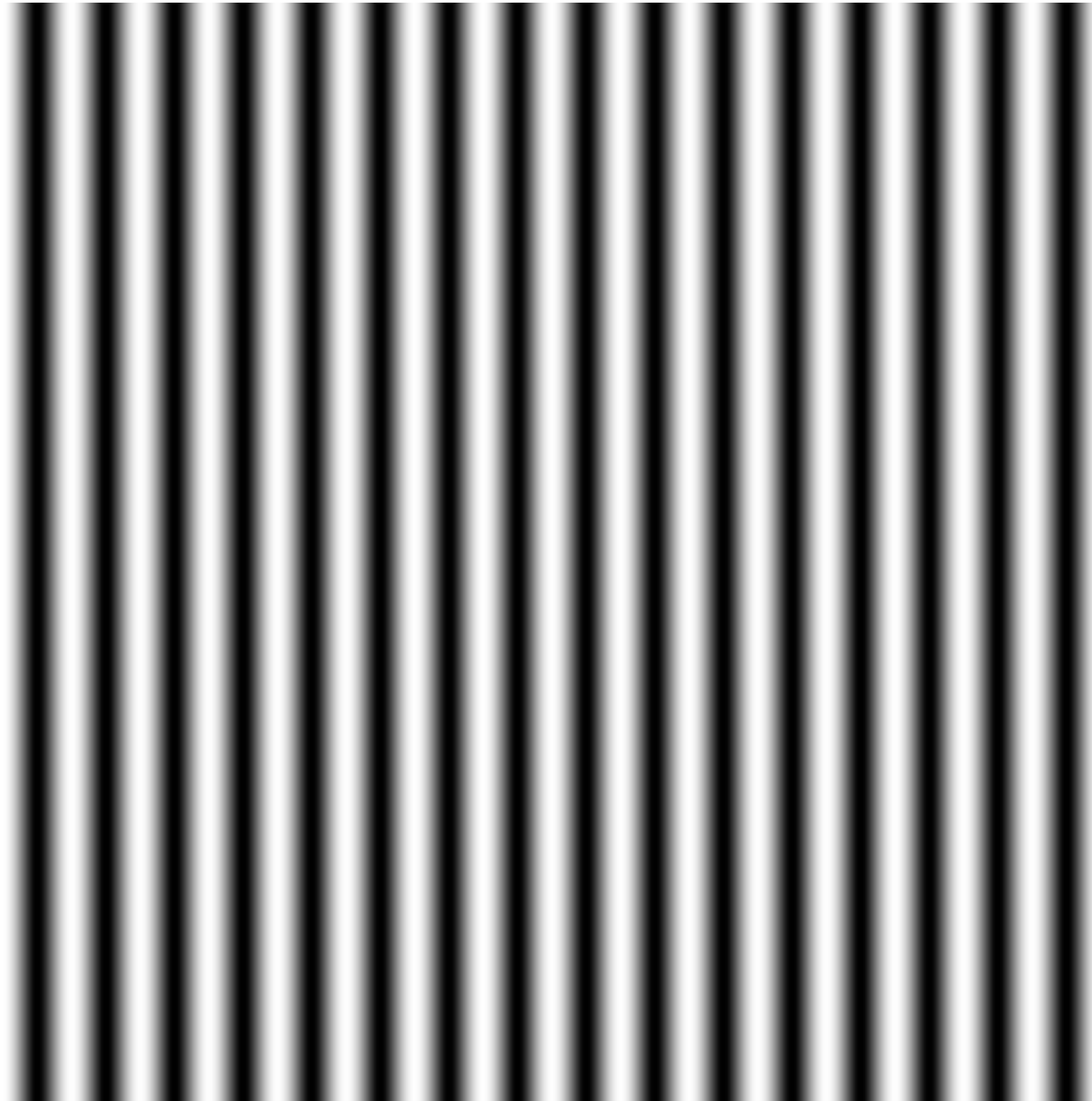


**Spatial domain**

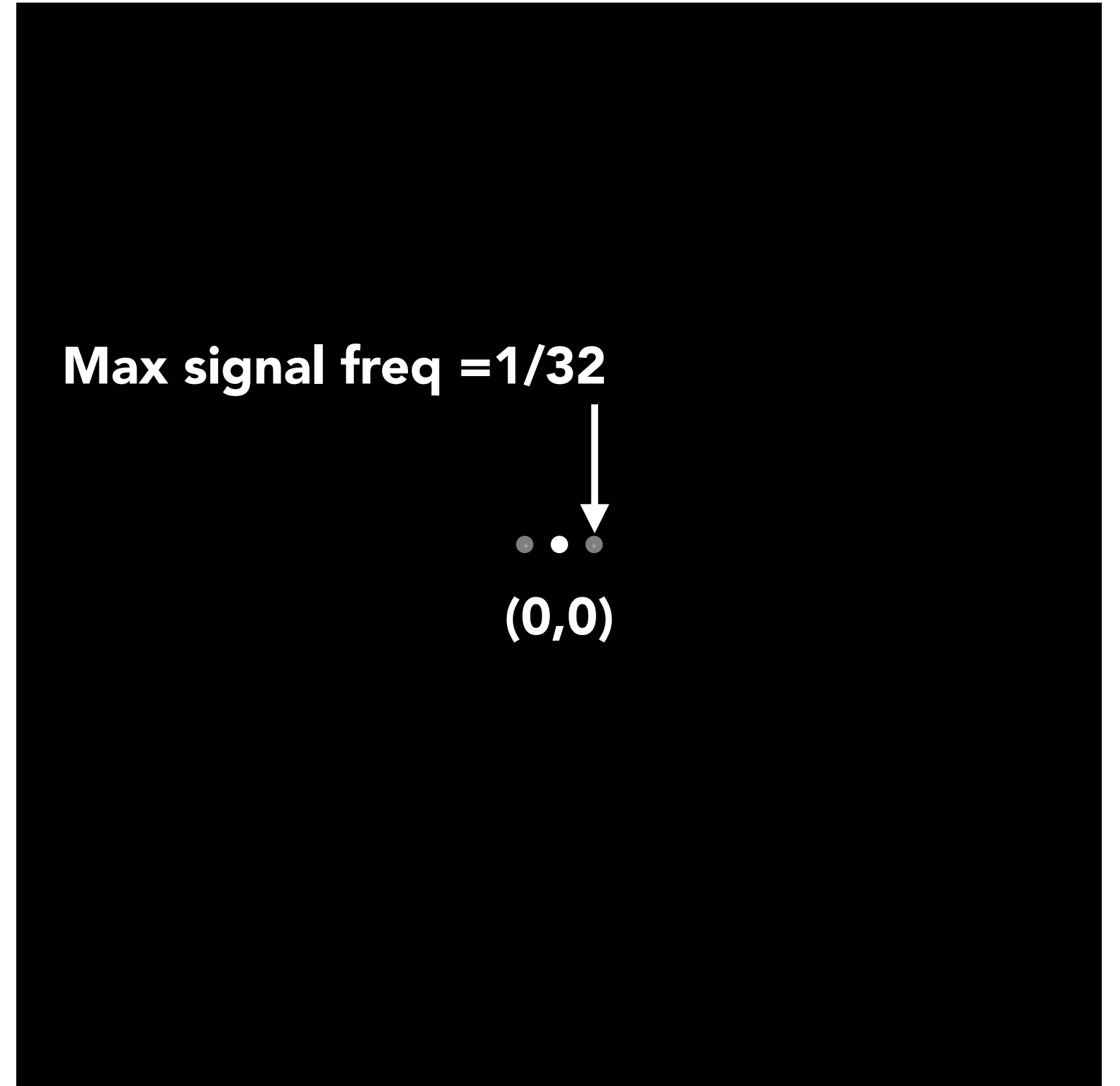


**Frequency domain**

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

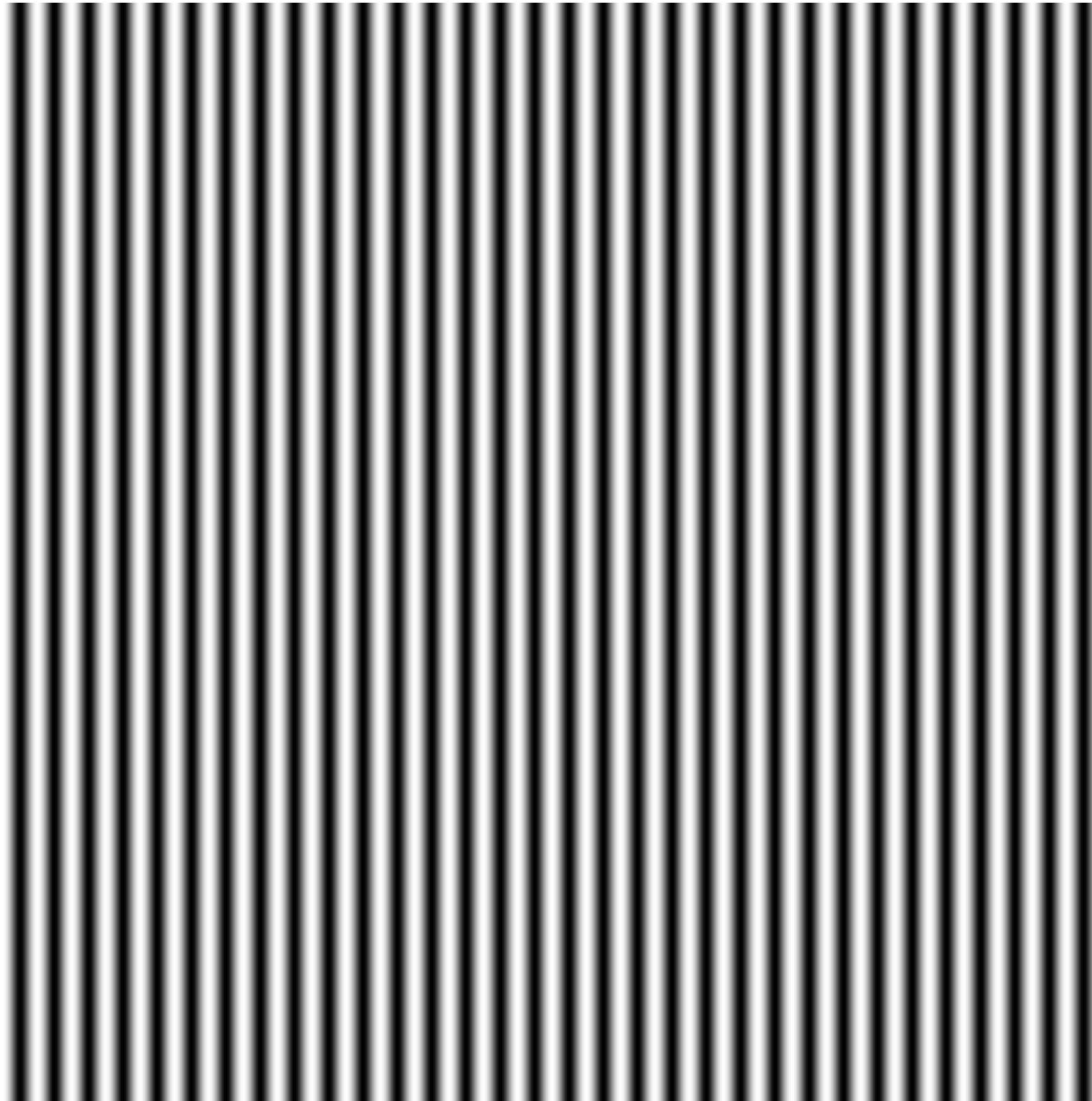


**Spatial domain**

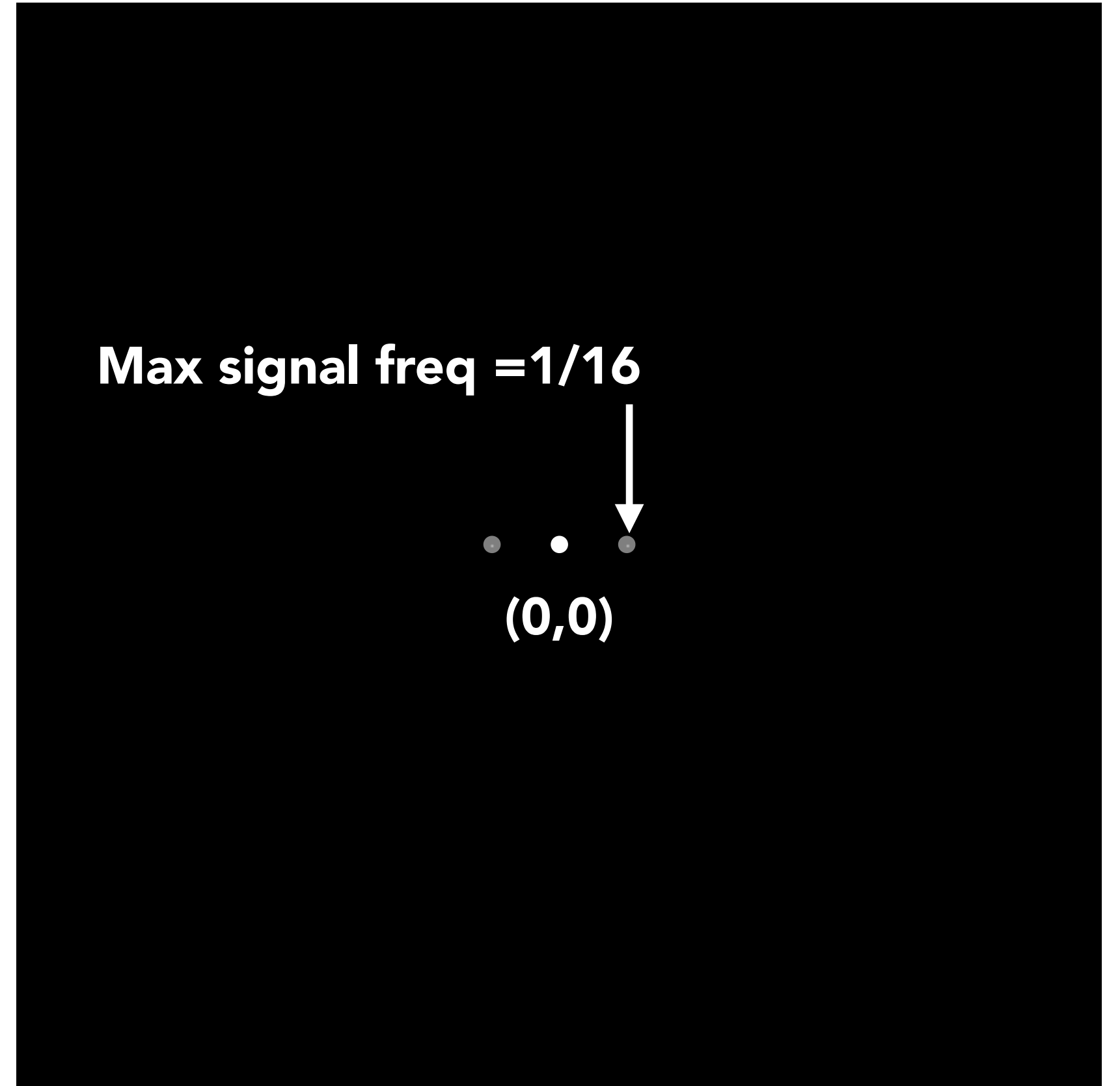


**Frequency domain**

$\sin(2\pi/16)x$  — frequency 1/16; 16 pixels per cycle

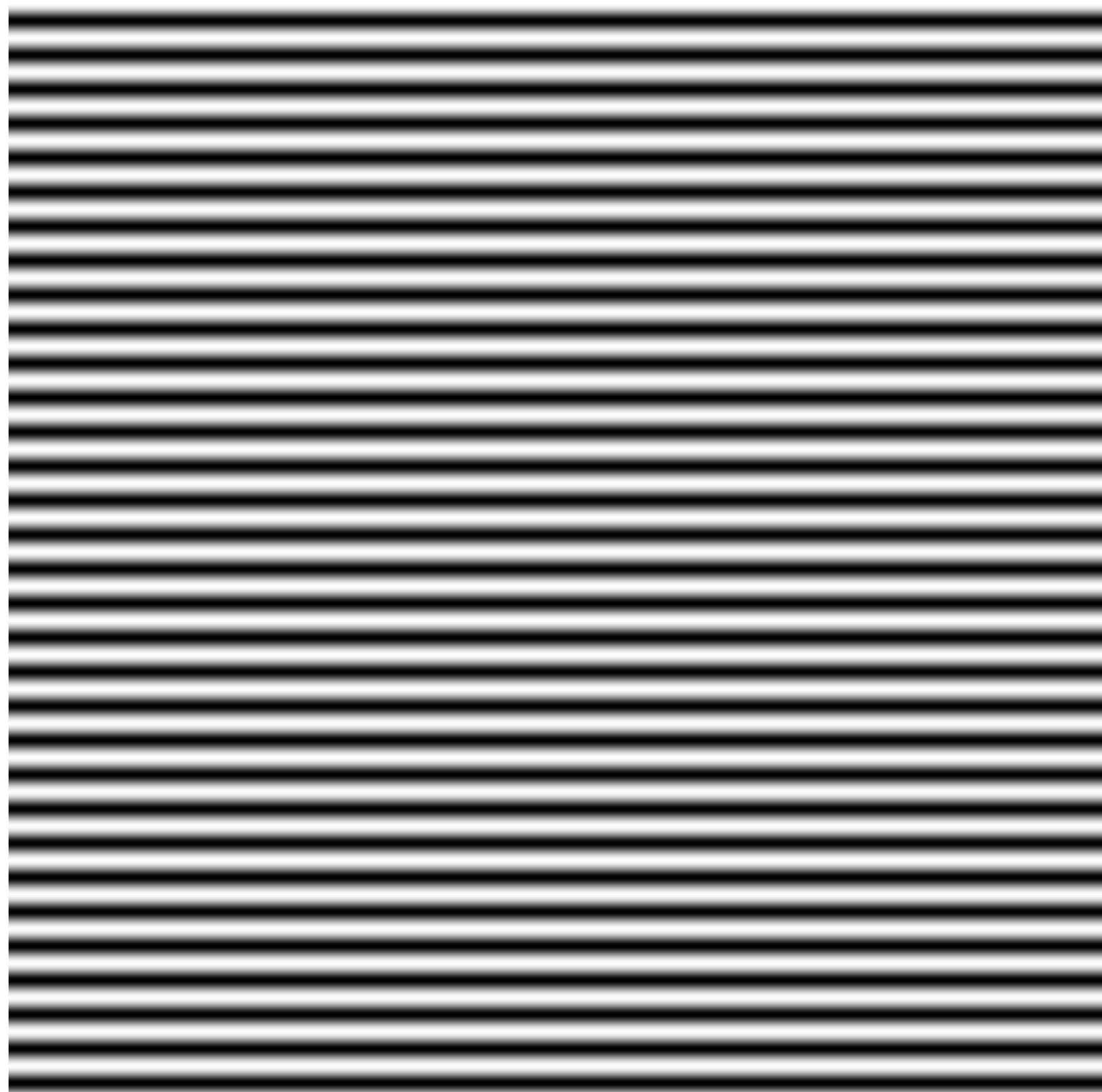


**Spatial domain**

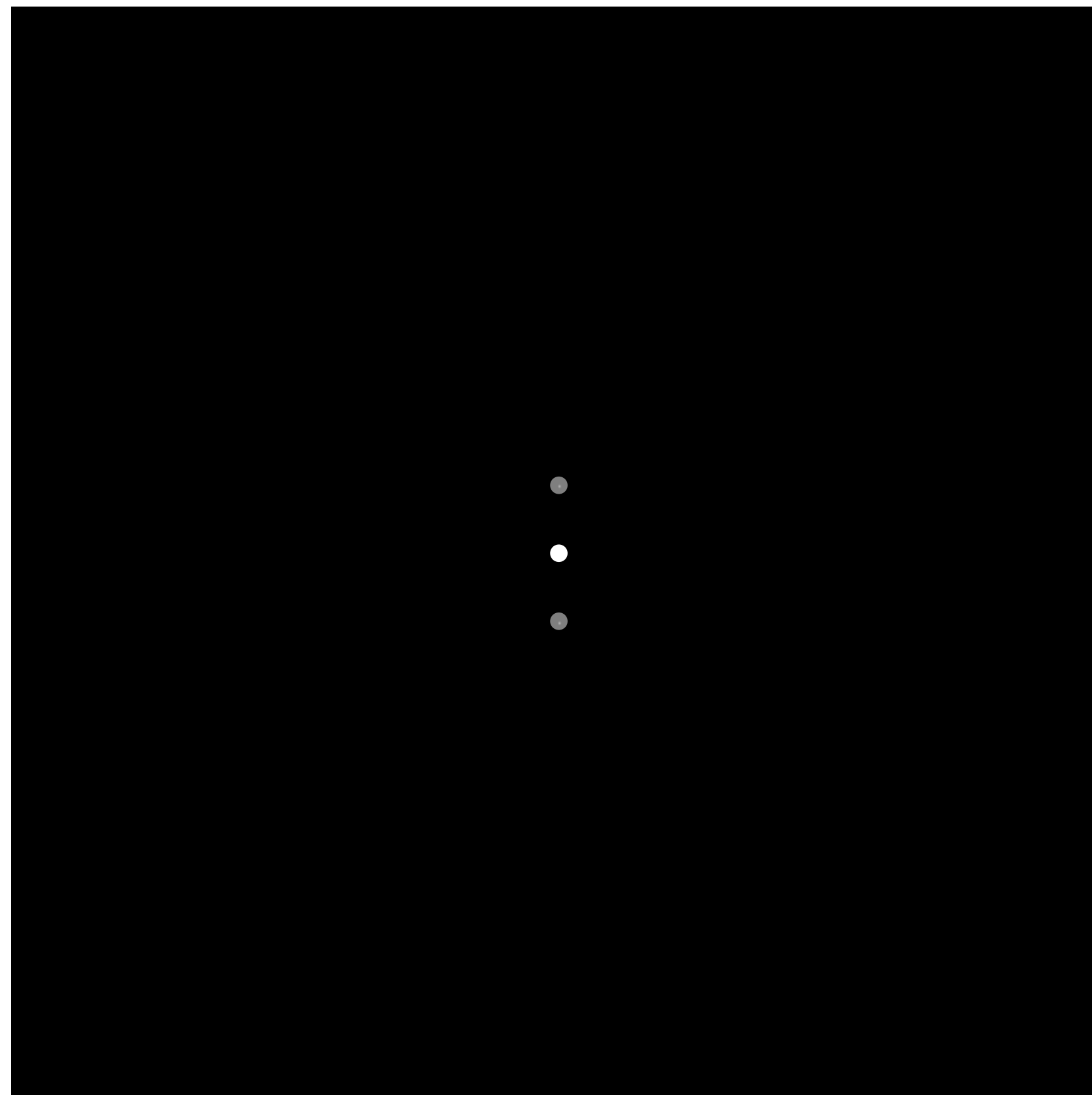


**Frequency domain**

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



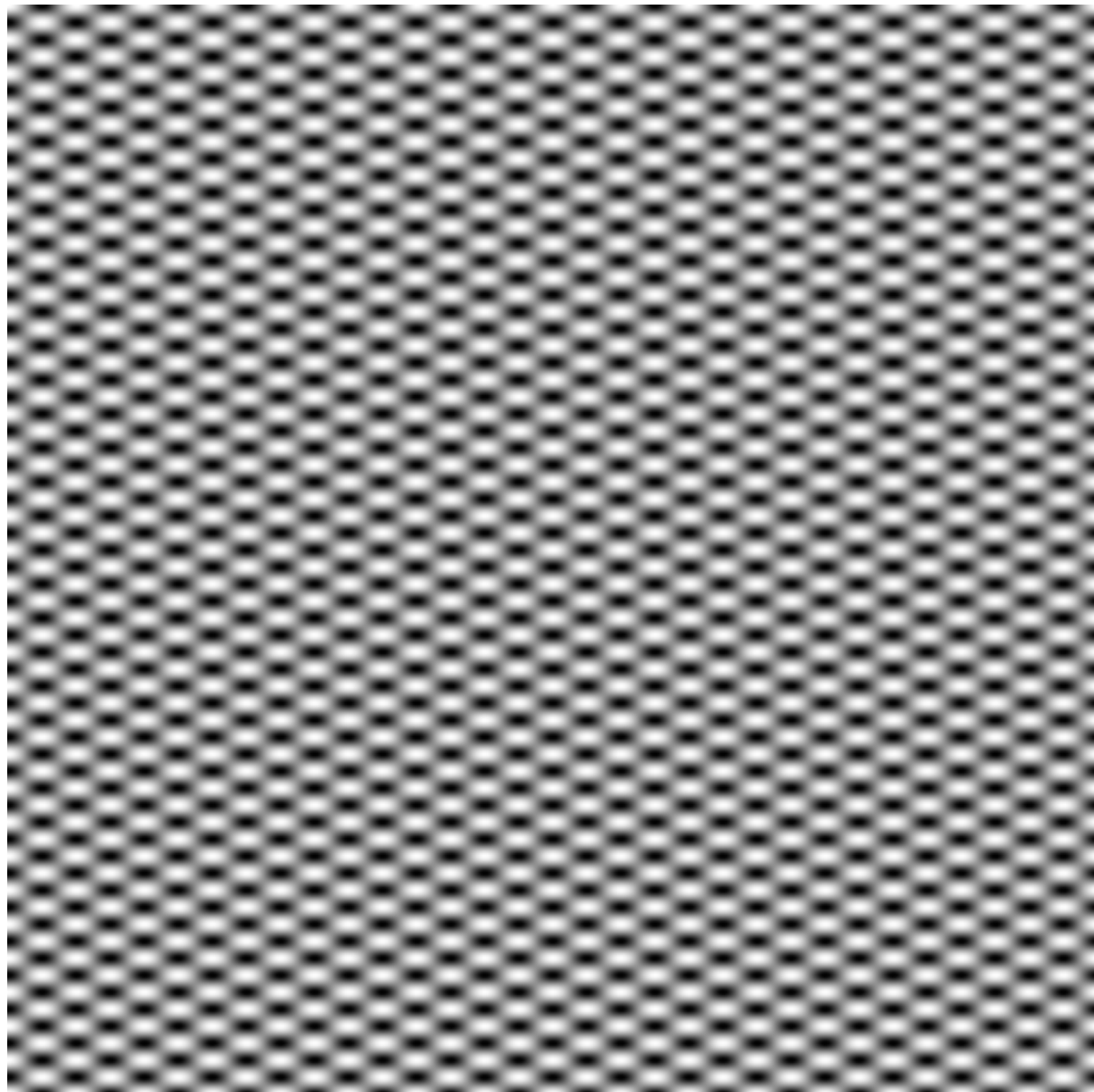
**Spatial domain**



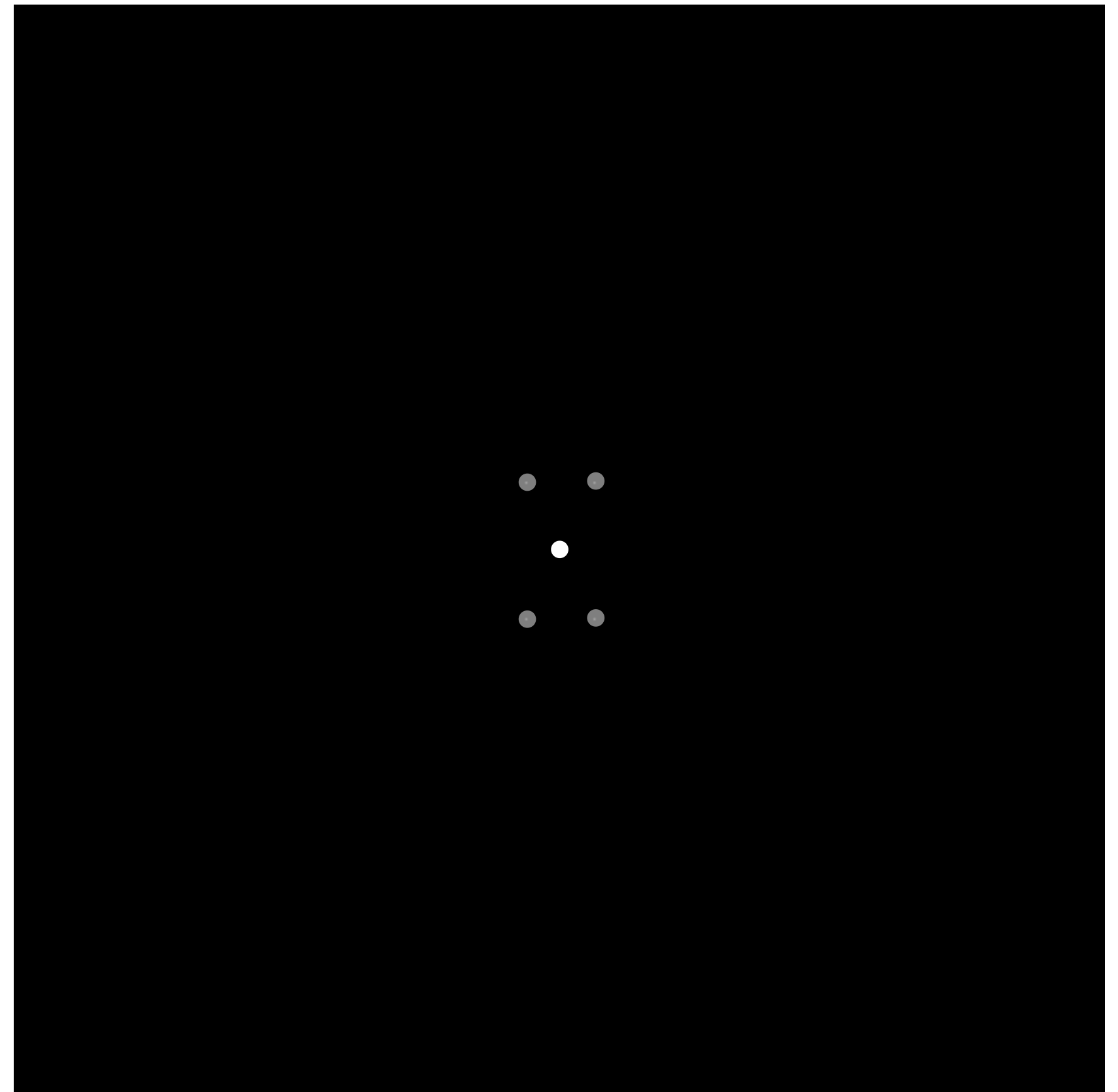
**Frequency domain**



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

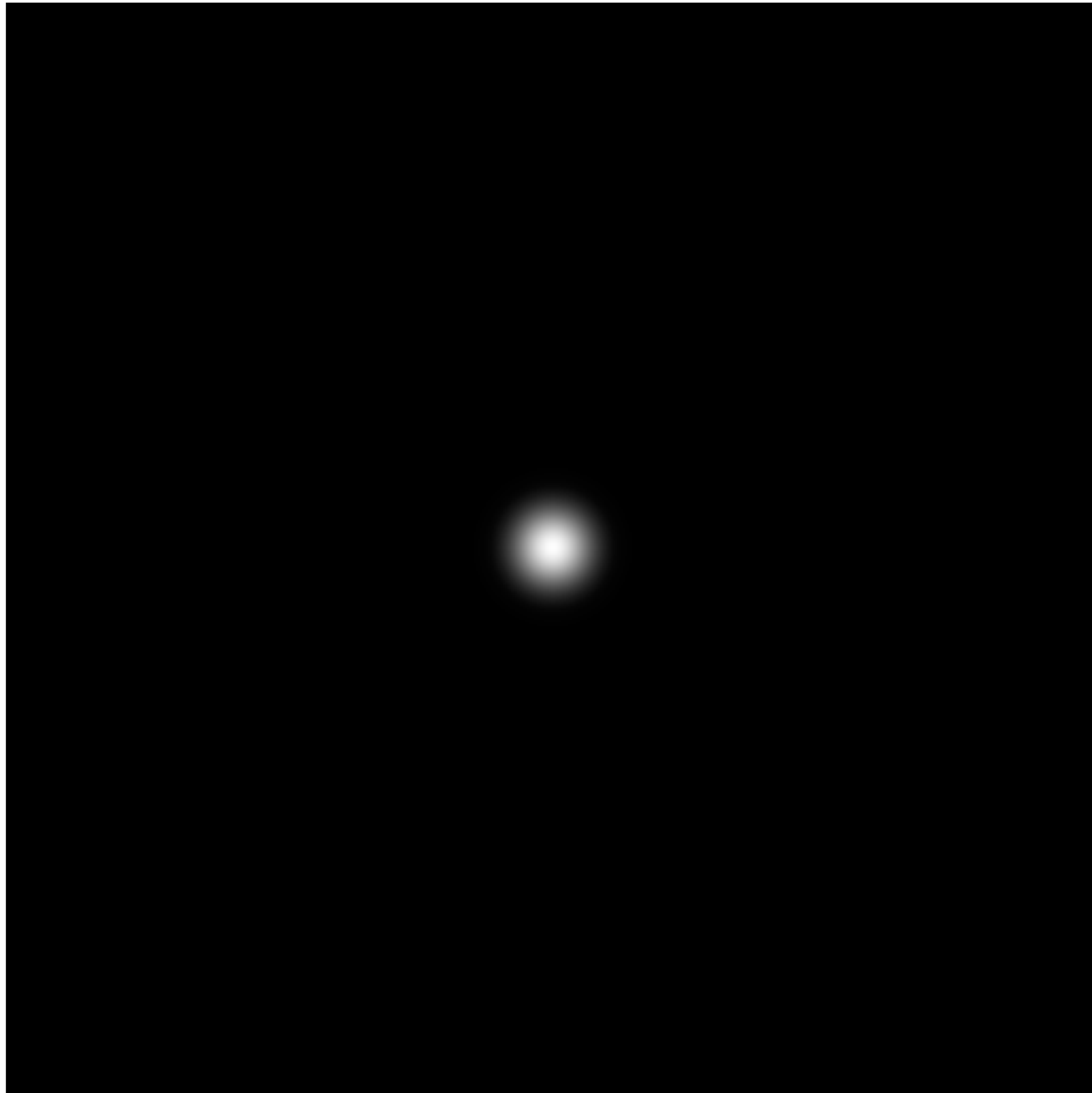


**Spatial domain**

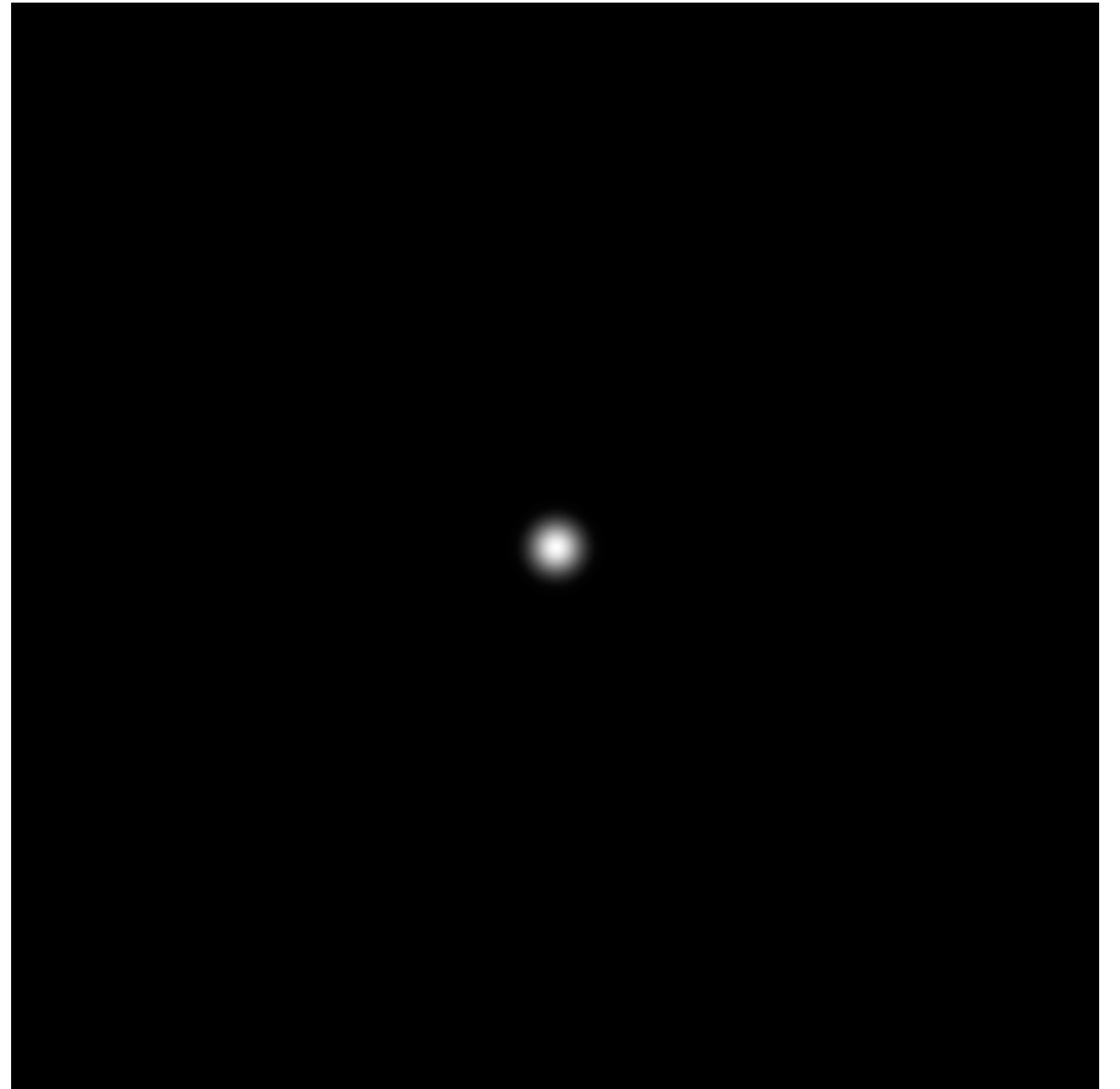


**Frequency domain**

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

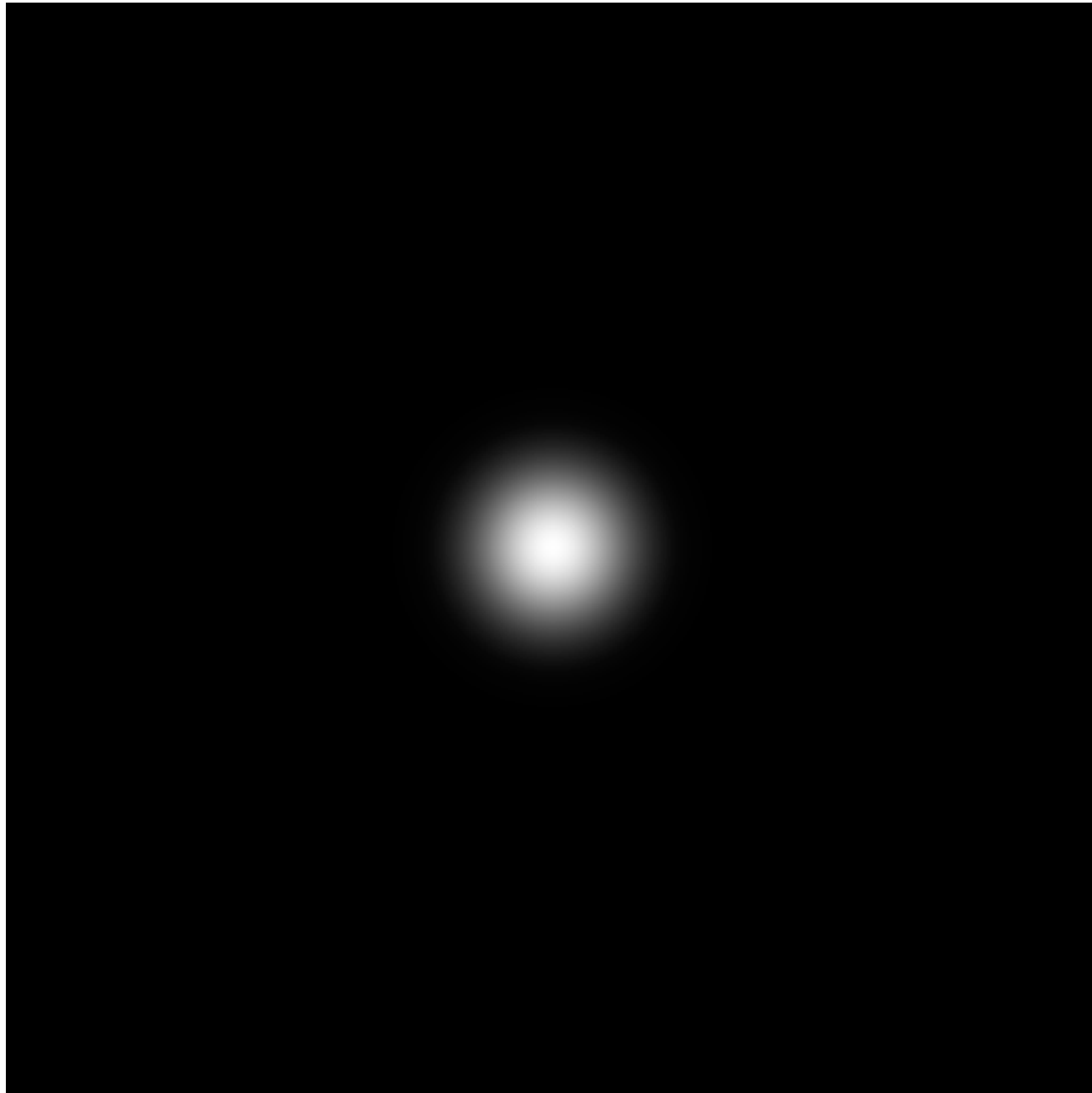


**Spatial domain**

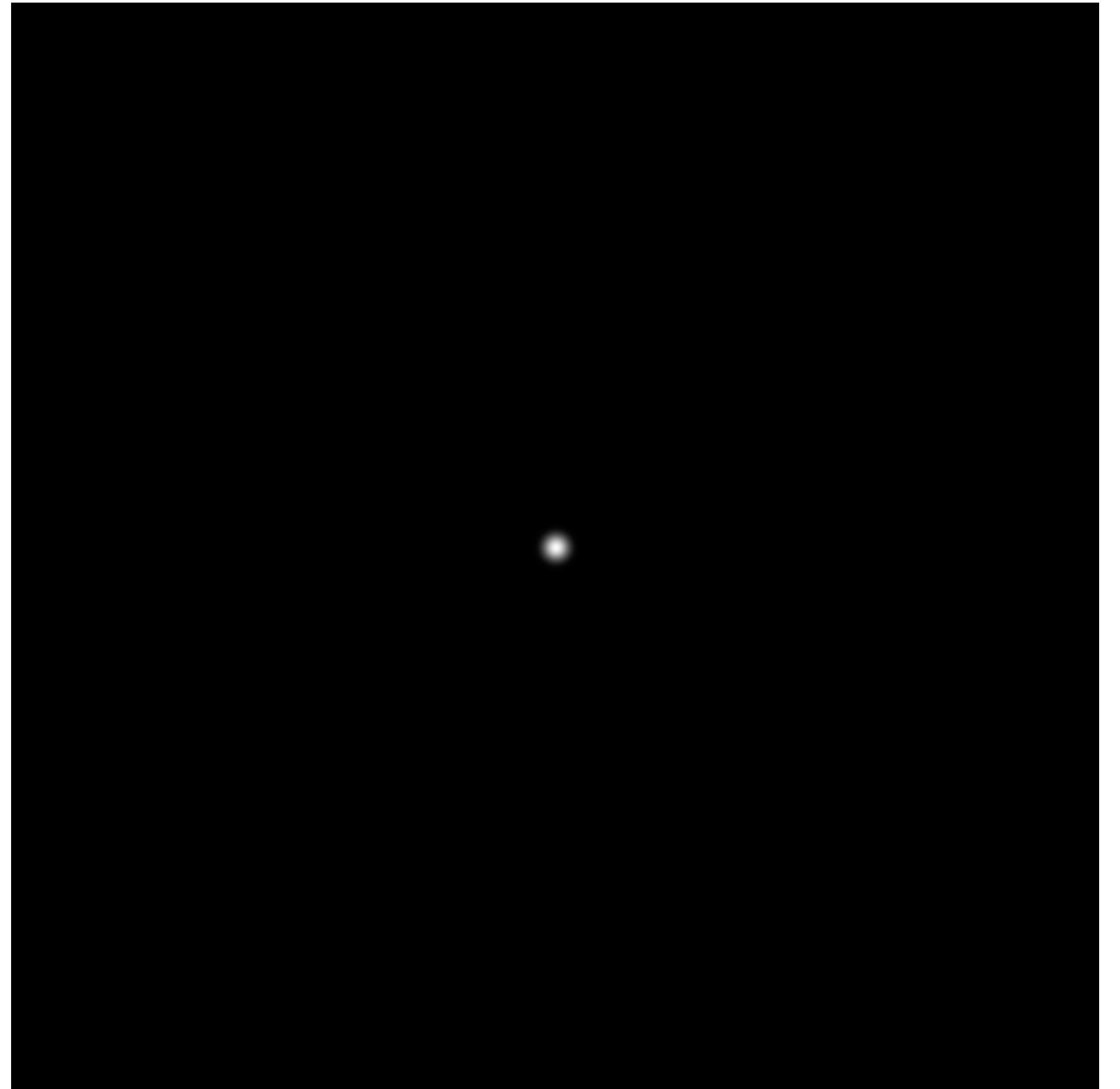


**Frequency domain**

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

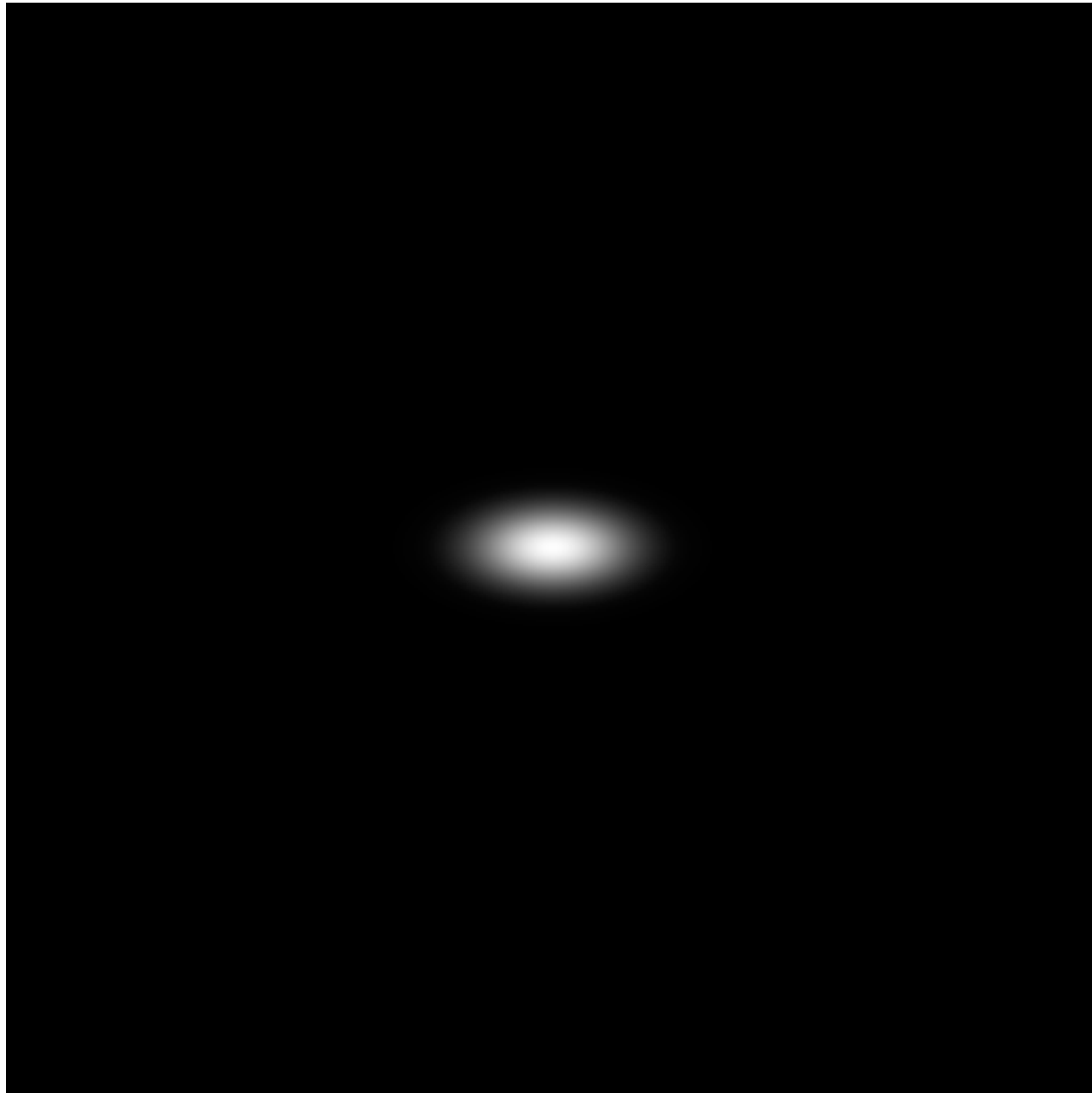


**Spatial domain**



**Frequency domain**

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



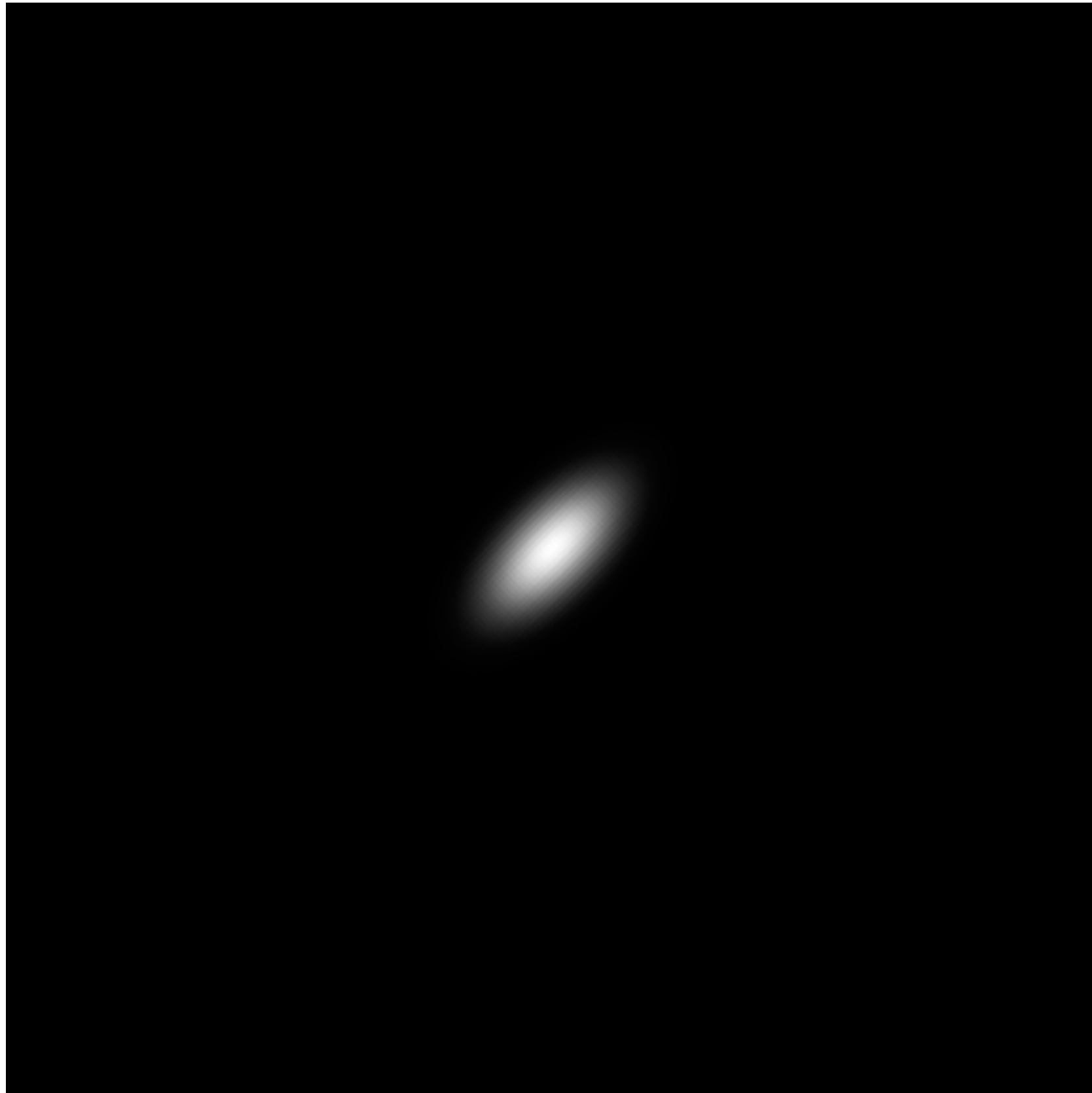
**Spatial domain**



**Frequency domain**

**Rotate 45**

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



**Spatial domain**



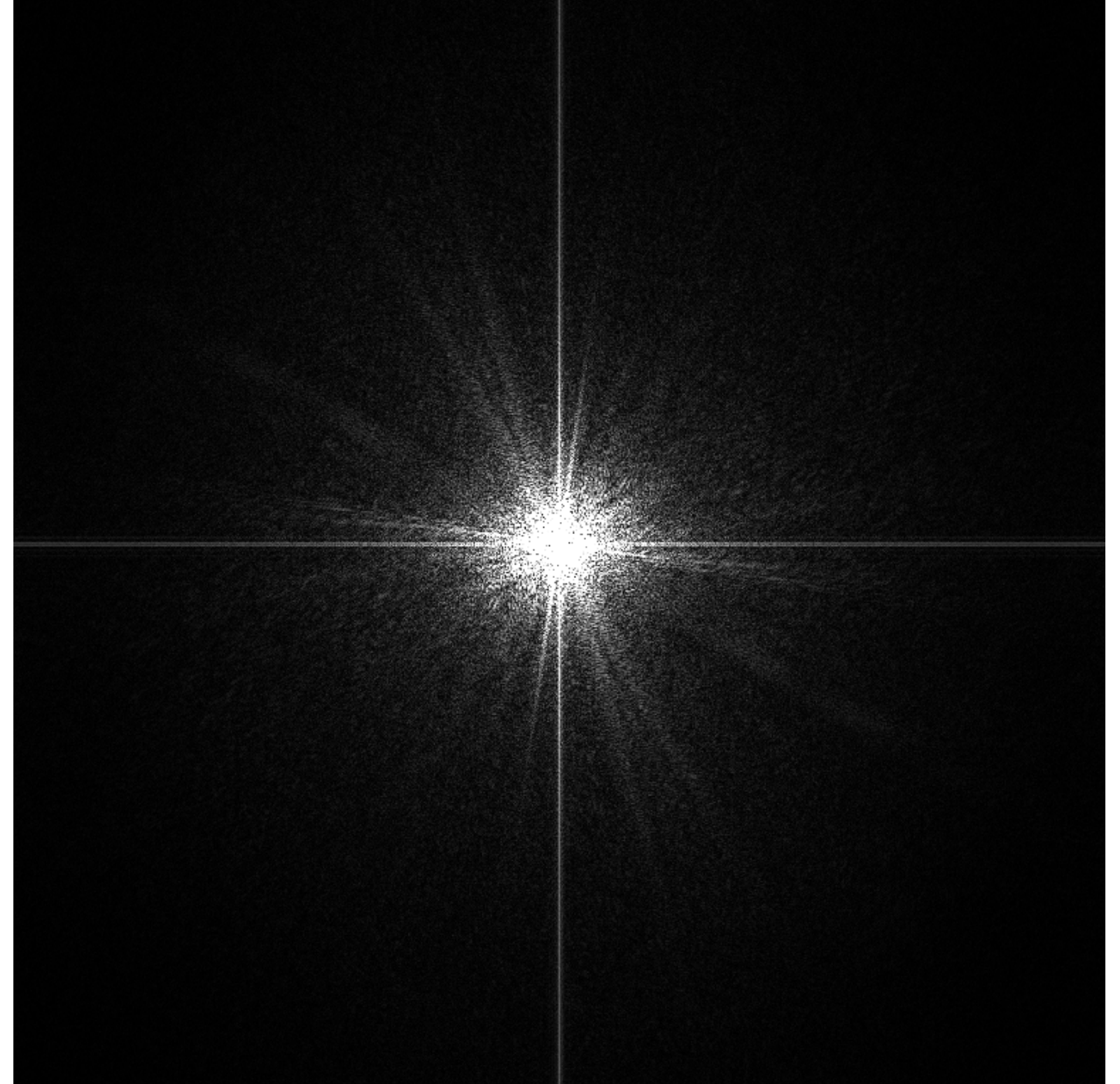
**Frequency domain**

# **Image filtering (in the frequency domain)**

# Visualizing the frequency content of images



**Spatial domain**

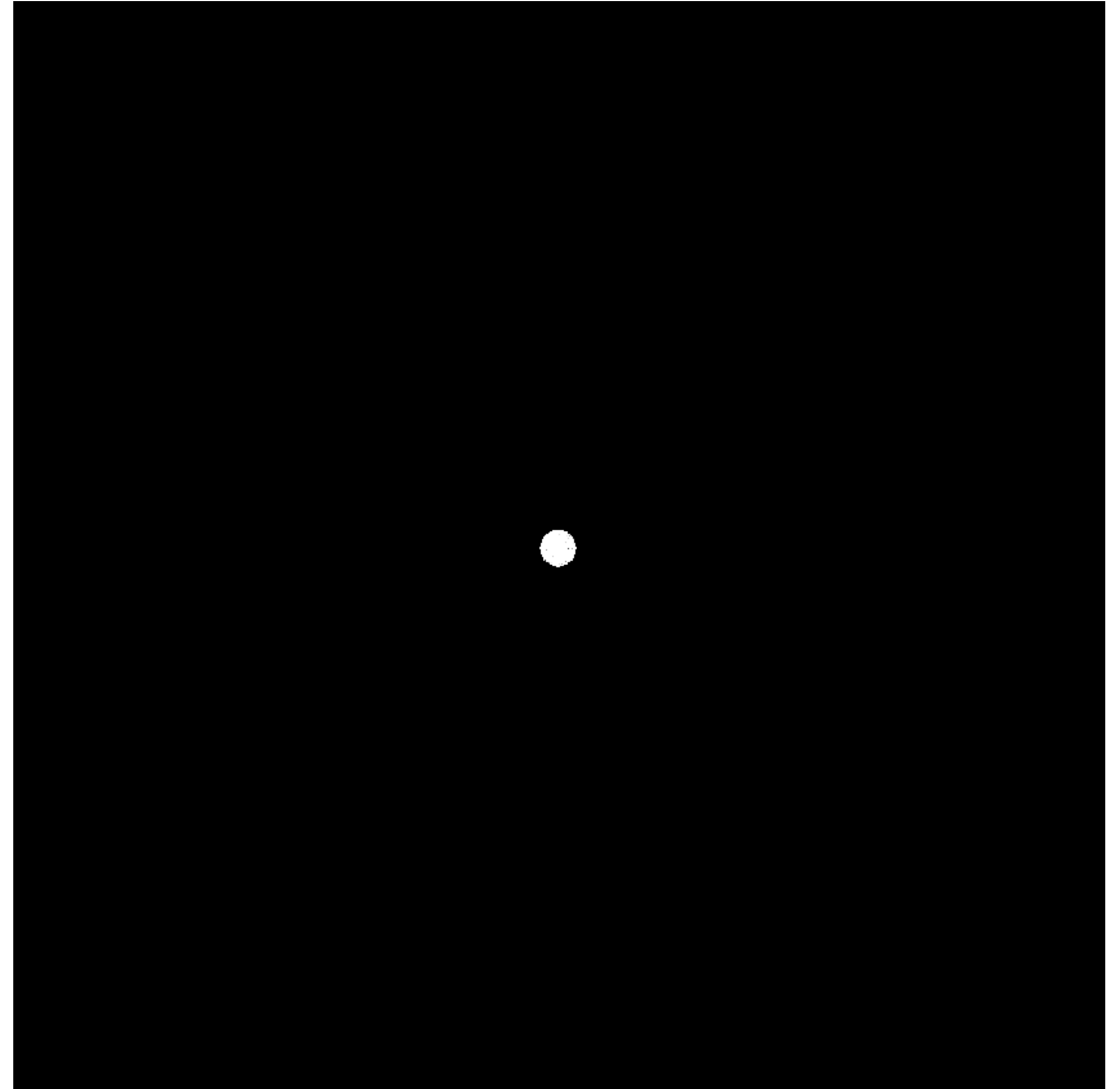


**Frequency domain**

# Low frequencies only (smooth gradients)



**Spatial domain**



**Frequency domain**

(after low-pass filter)

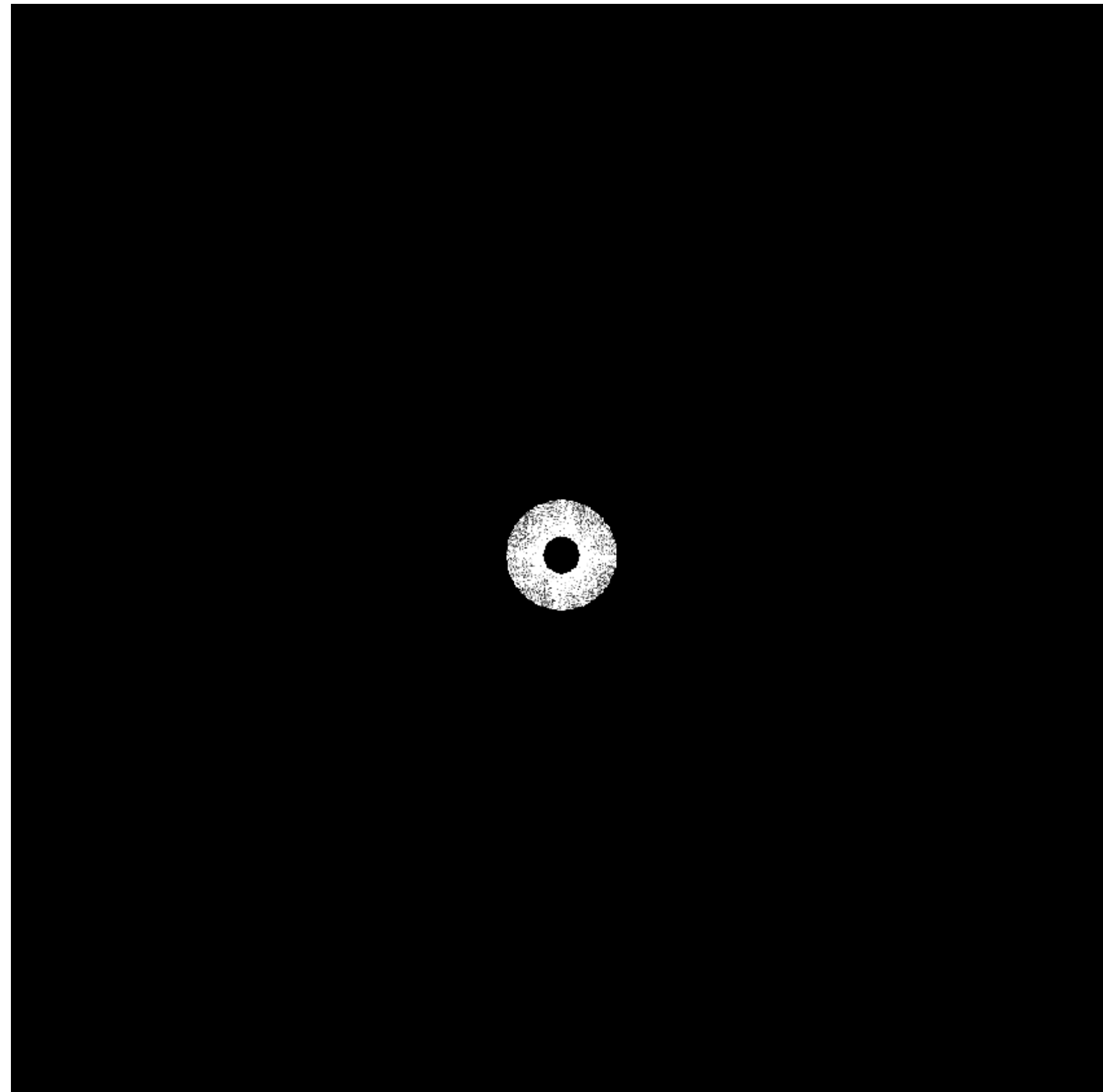
All frequencies above cutoff have 0 magnitude



# Mid-range frequencies



**Spatial domain**

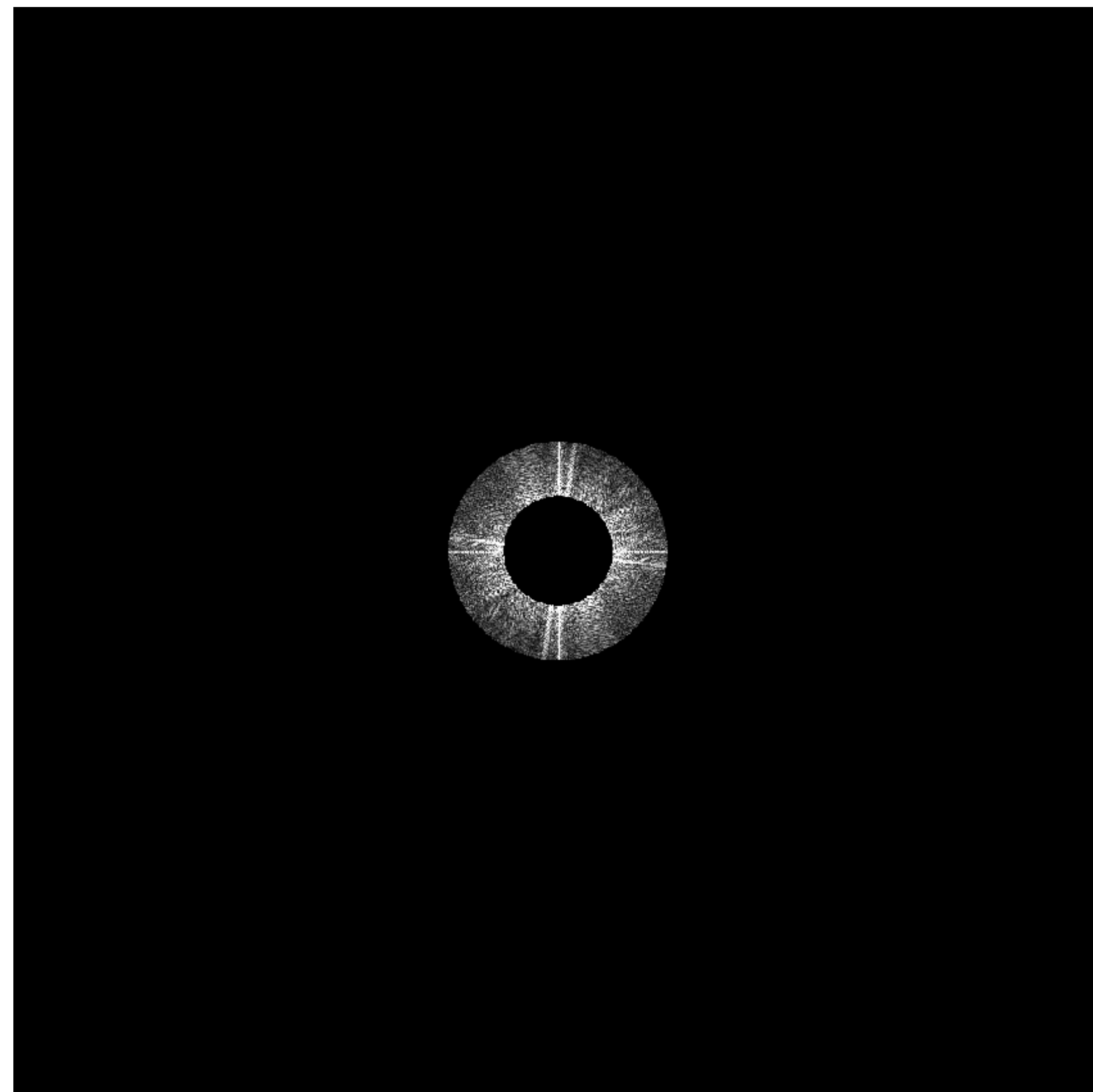


**Frequency domain**  
(after band-pass filter)

# Mid-range frequencies



**Spatial domain**

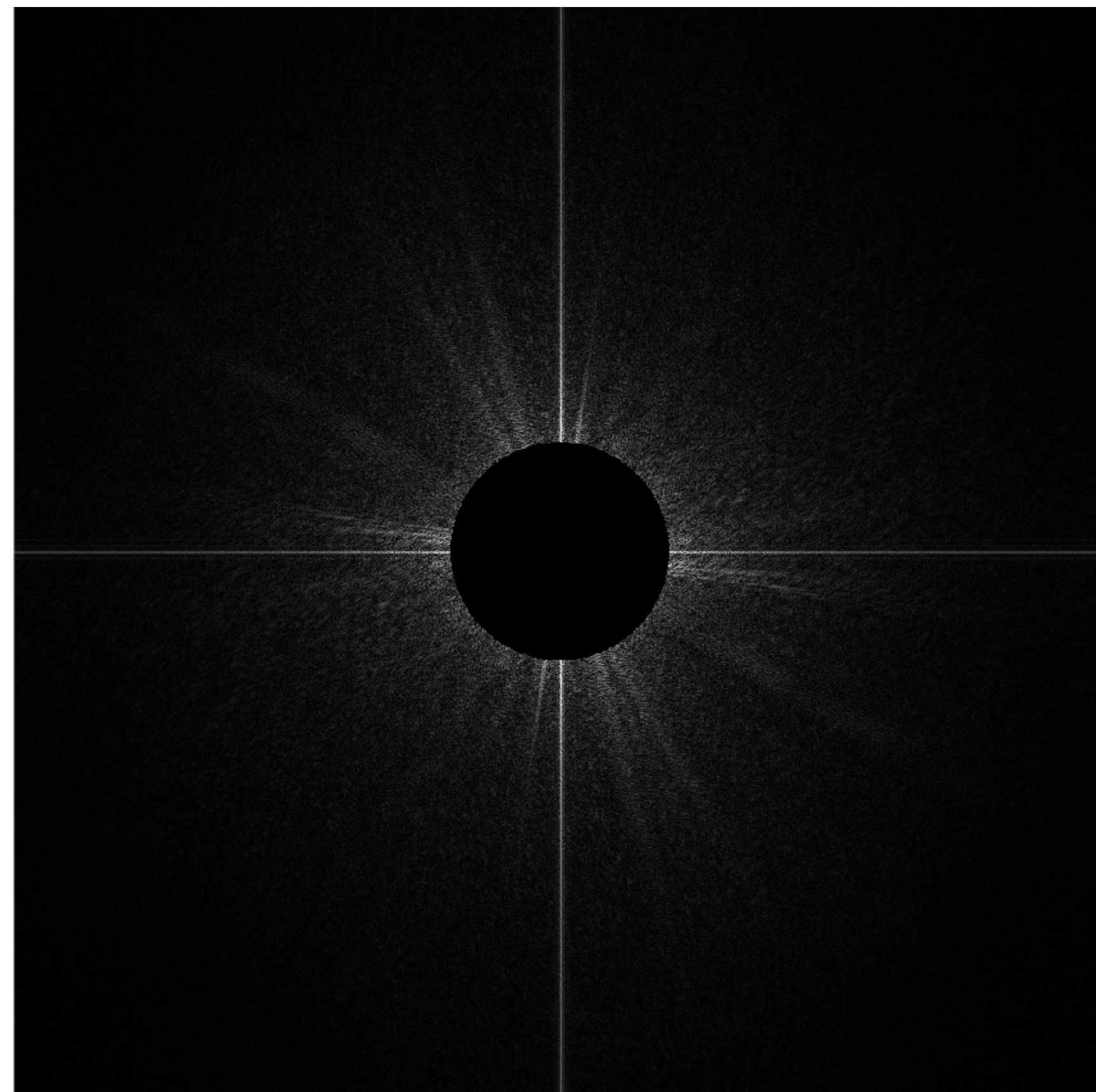


**Frequency domain**  
(after band-pass filter)

# High frequencies (edges)

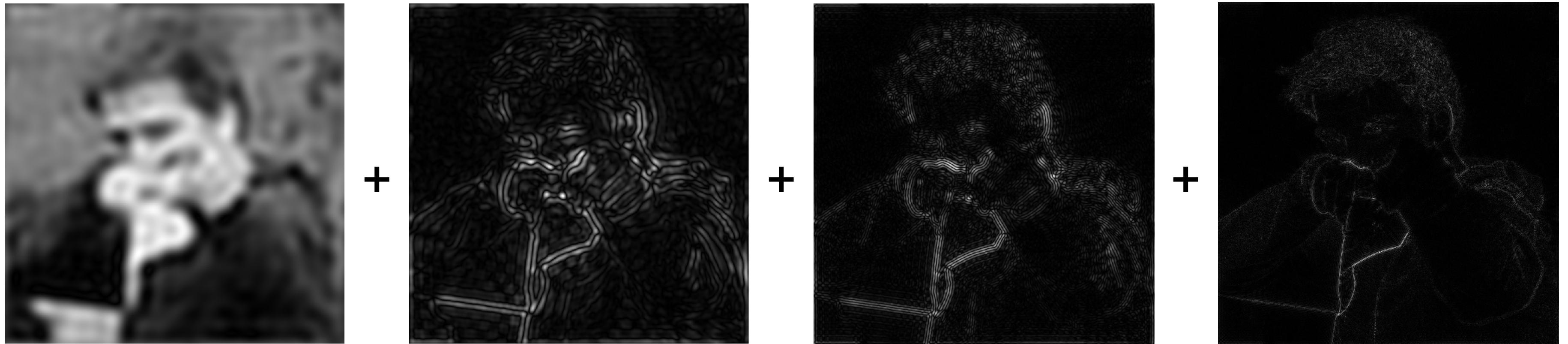


**Spatial domain**  
(strongest edges)



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

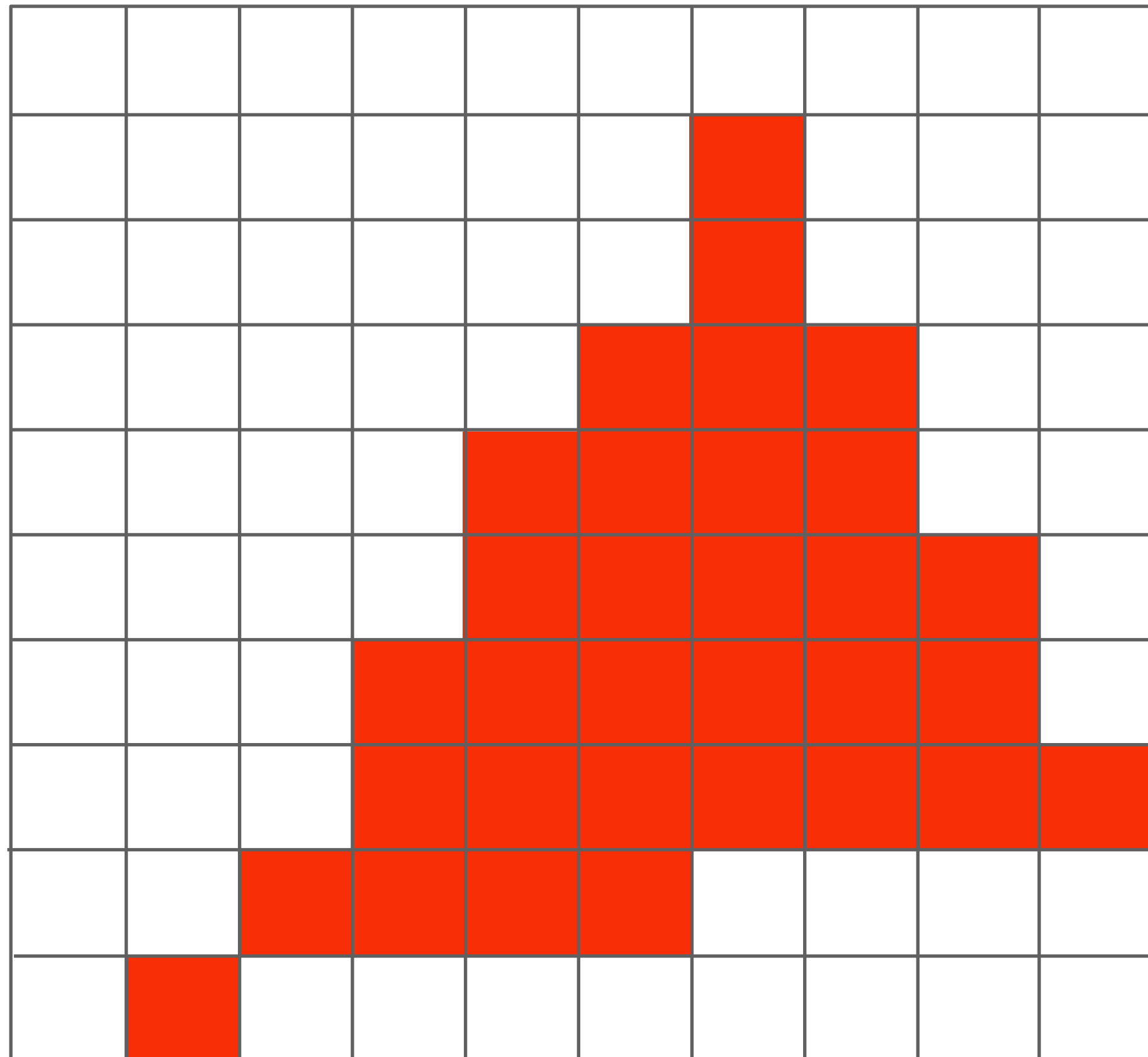
# An image as a sum of its frequency components



=

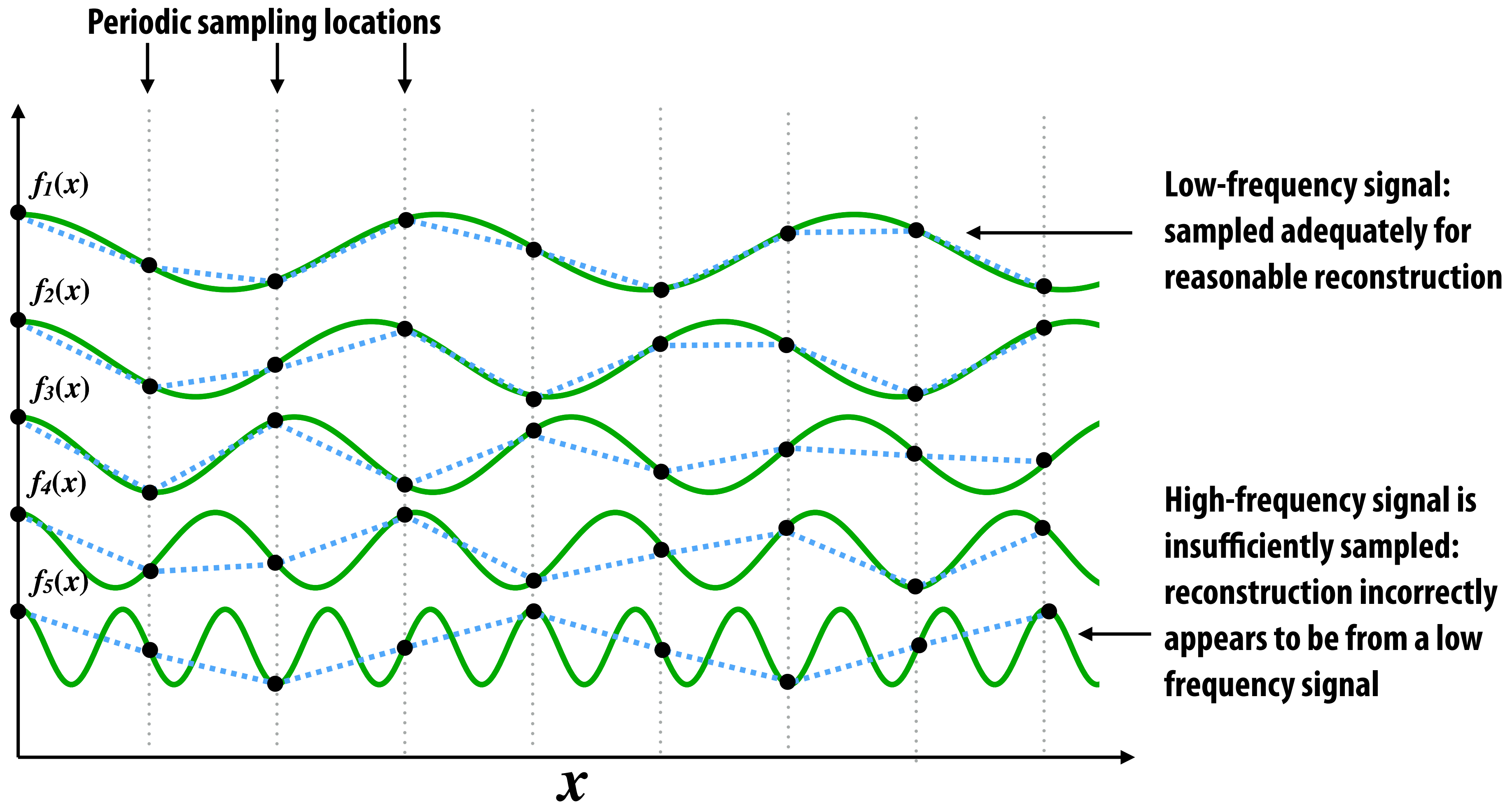


# Back to our problem of artifacts in images

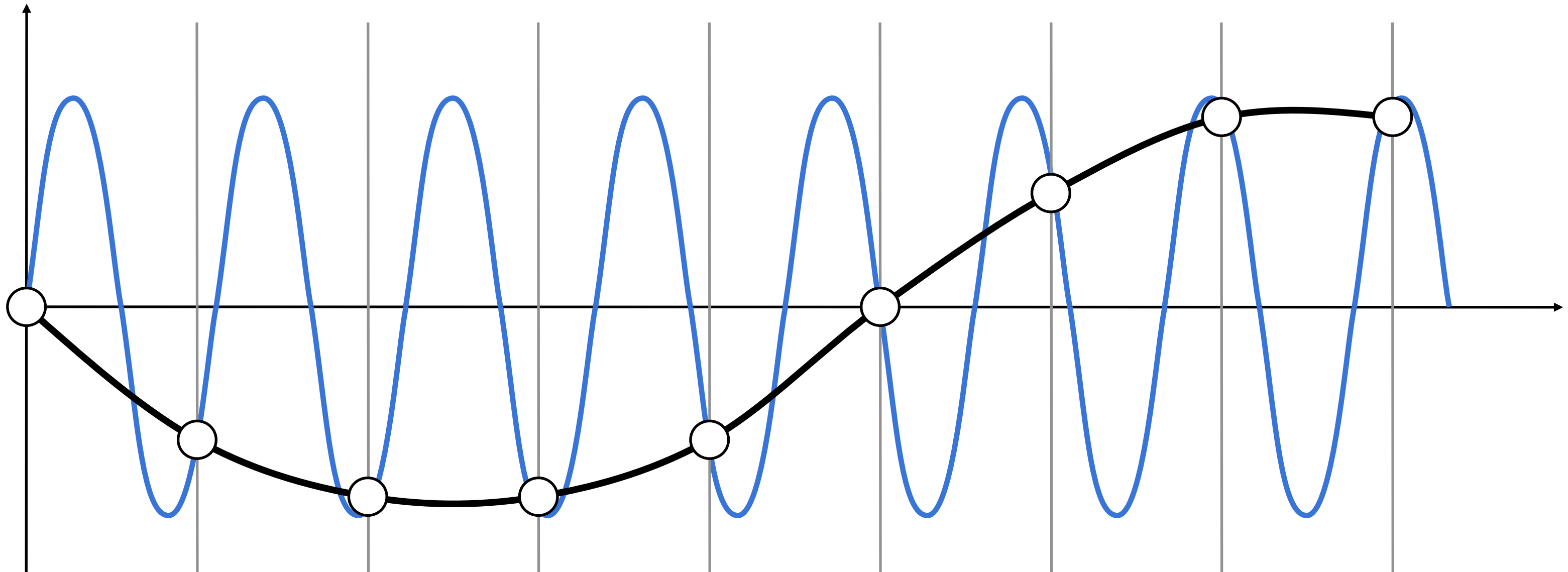


**Jaggies!**

# Higher frequencies need denser sampling



# 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"**

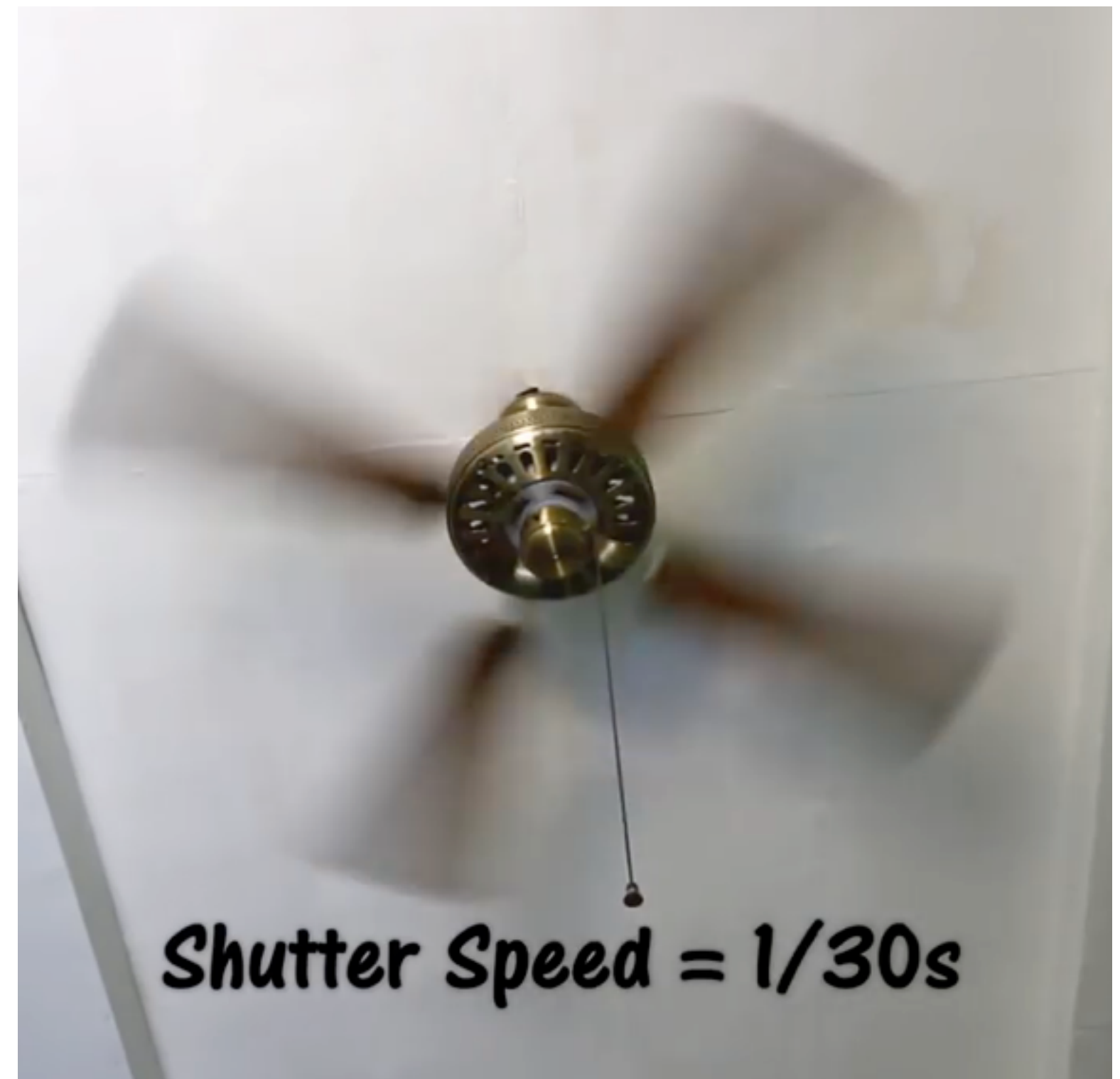
**Anti-aliasing idea: filter out high frequencies before sampling**



# Video: point vs antialiased sampling



**Point in time**



**Motion blurred**

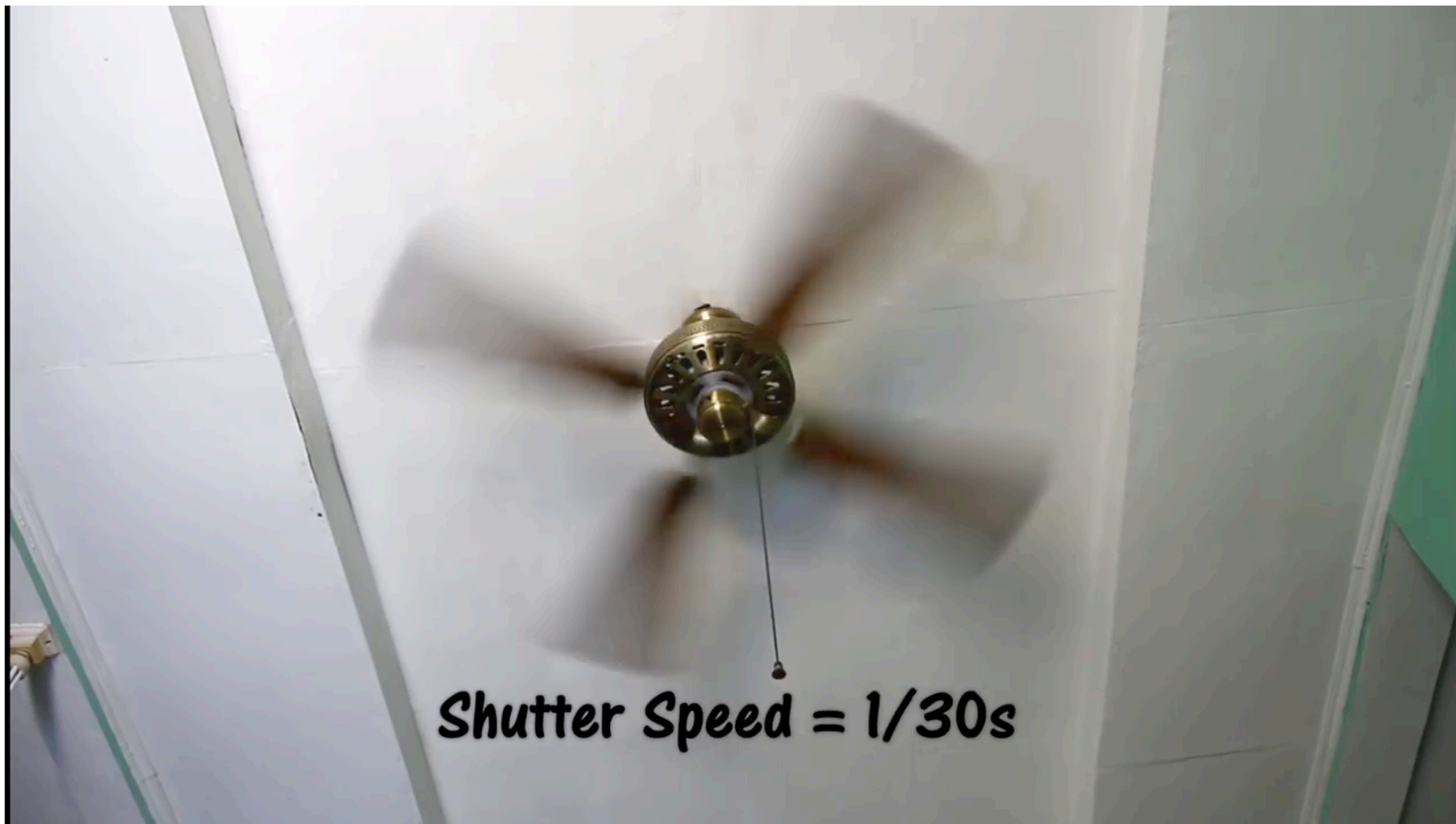
# Video: point sampling in time



Credit: Aris & cams youtube, <https://youtu.be/NoWwxTktoFs>

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

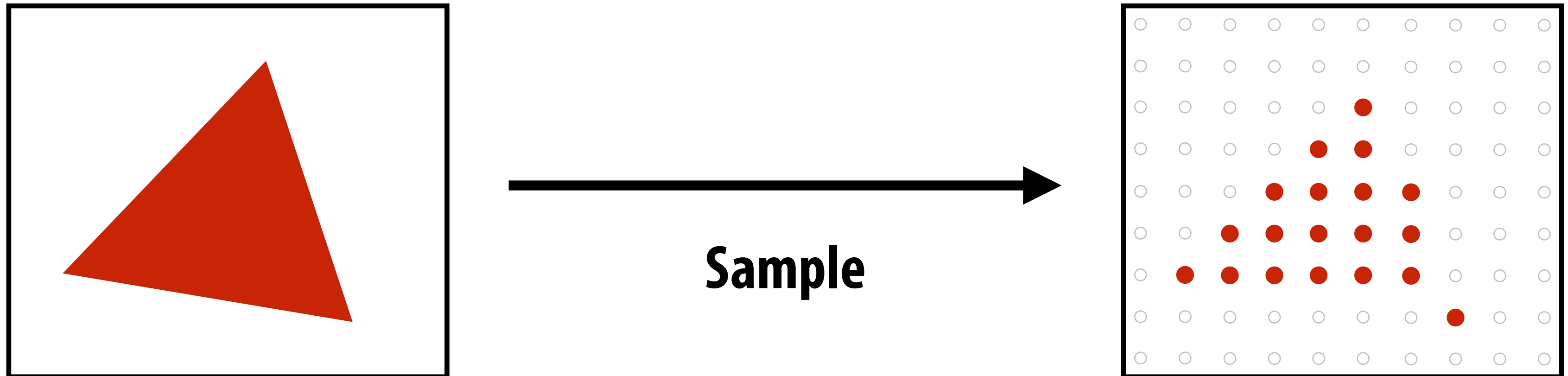
# Video: motion-blurred sampling



Credit: Aris & cams youtube, <https://youtu.be/NoWwxTktoFs>

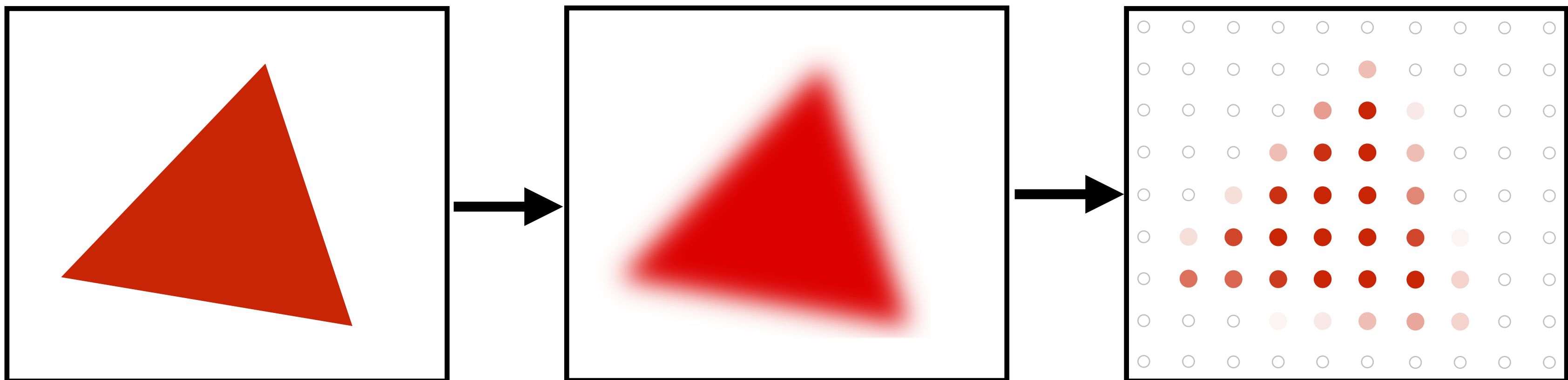
**30 fps video.  $1/30$  second exposure is motion-blurred in time, reduces aliasing.**

# Rasterization: point sampling in 2D space



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

# Rasterization: anti-aliased sampling



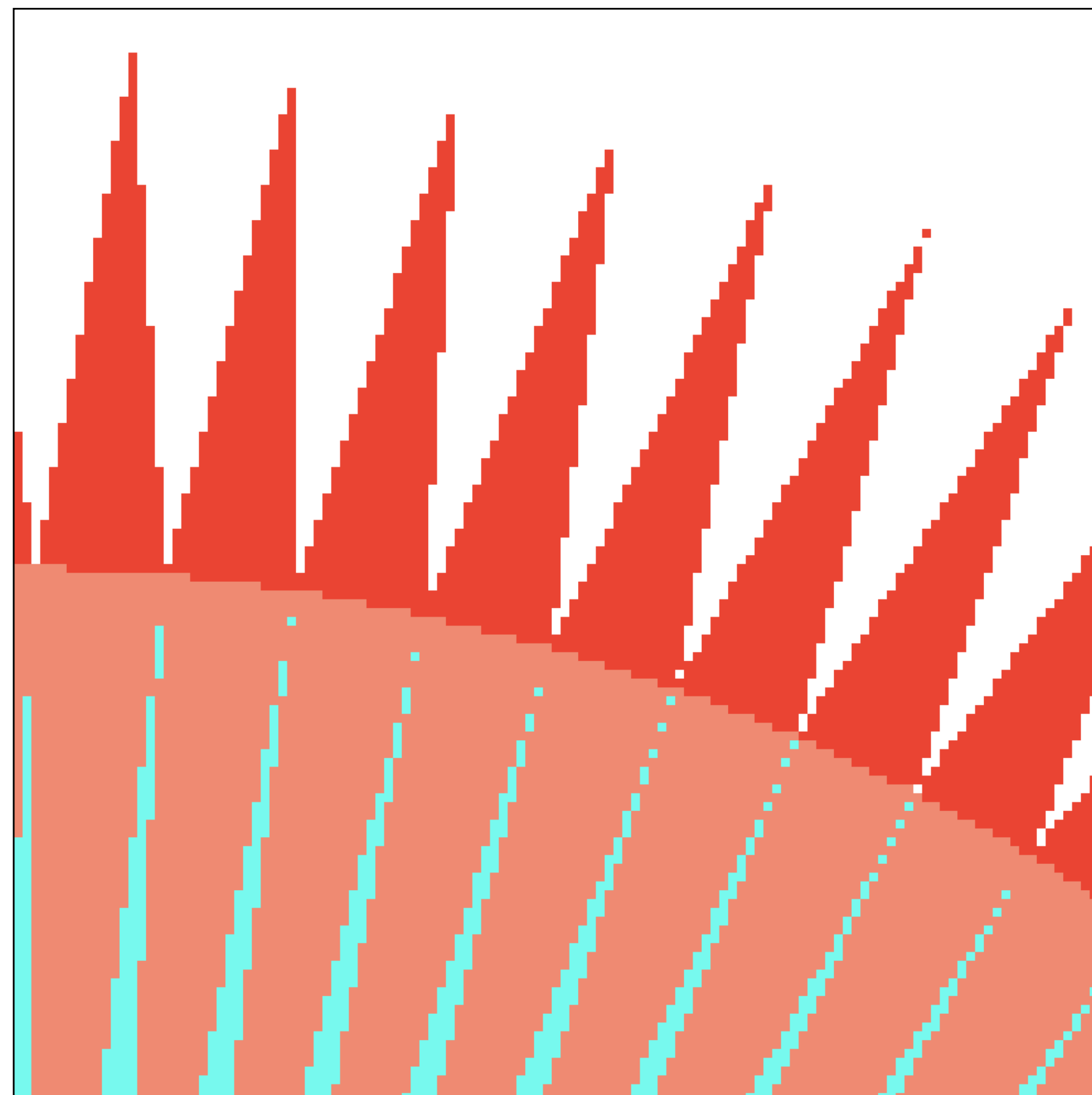
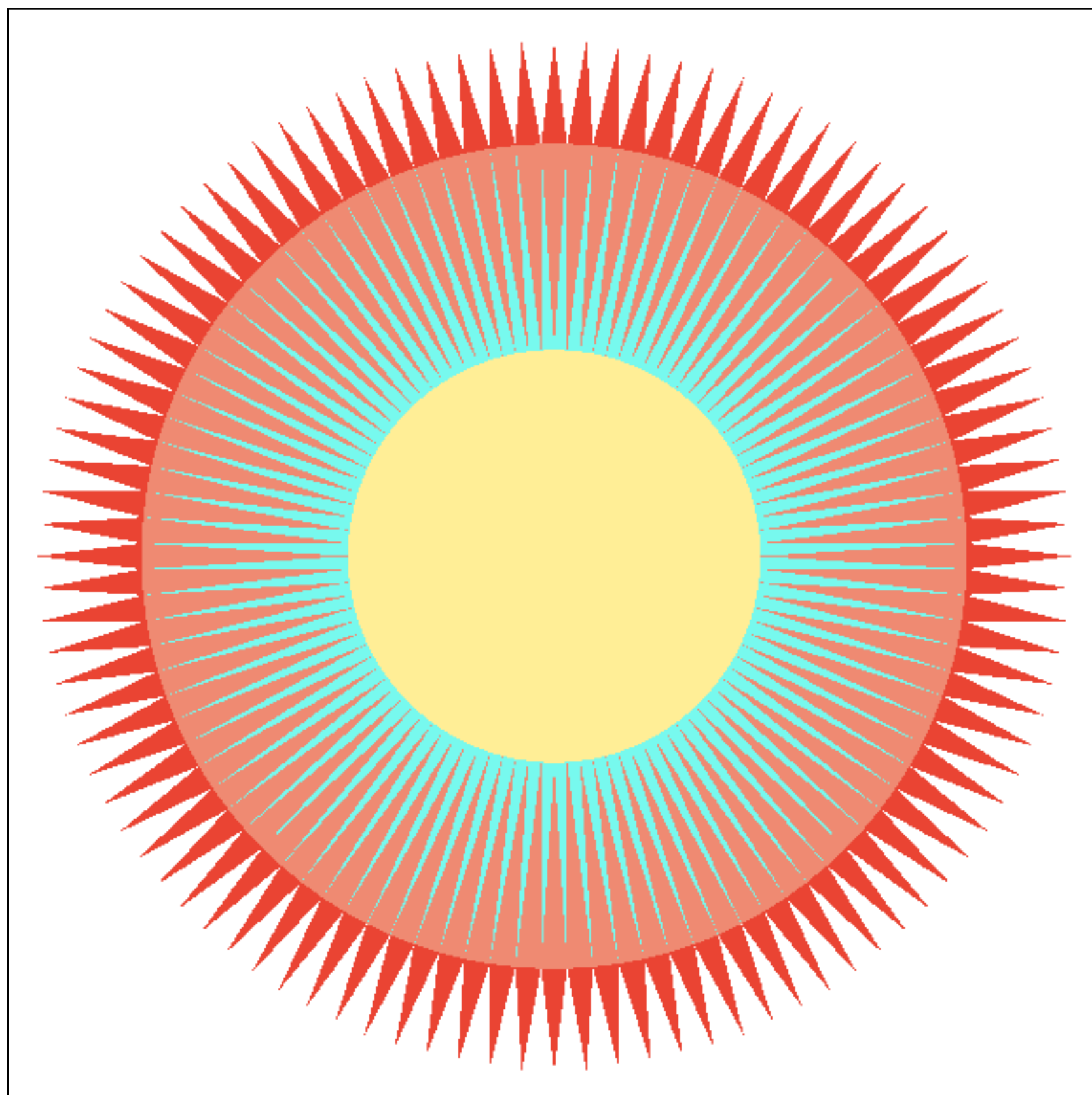
**Pre-filter**

(remove frequencies above Nyquist)

**Sample**

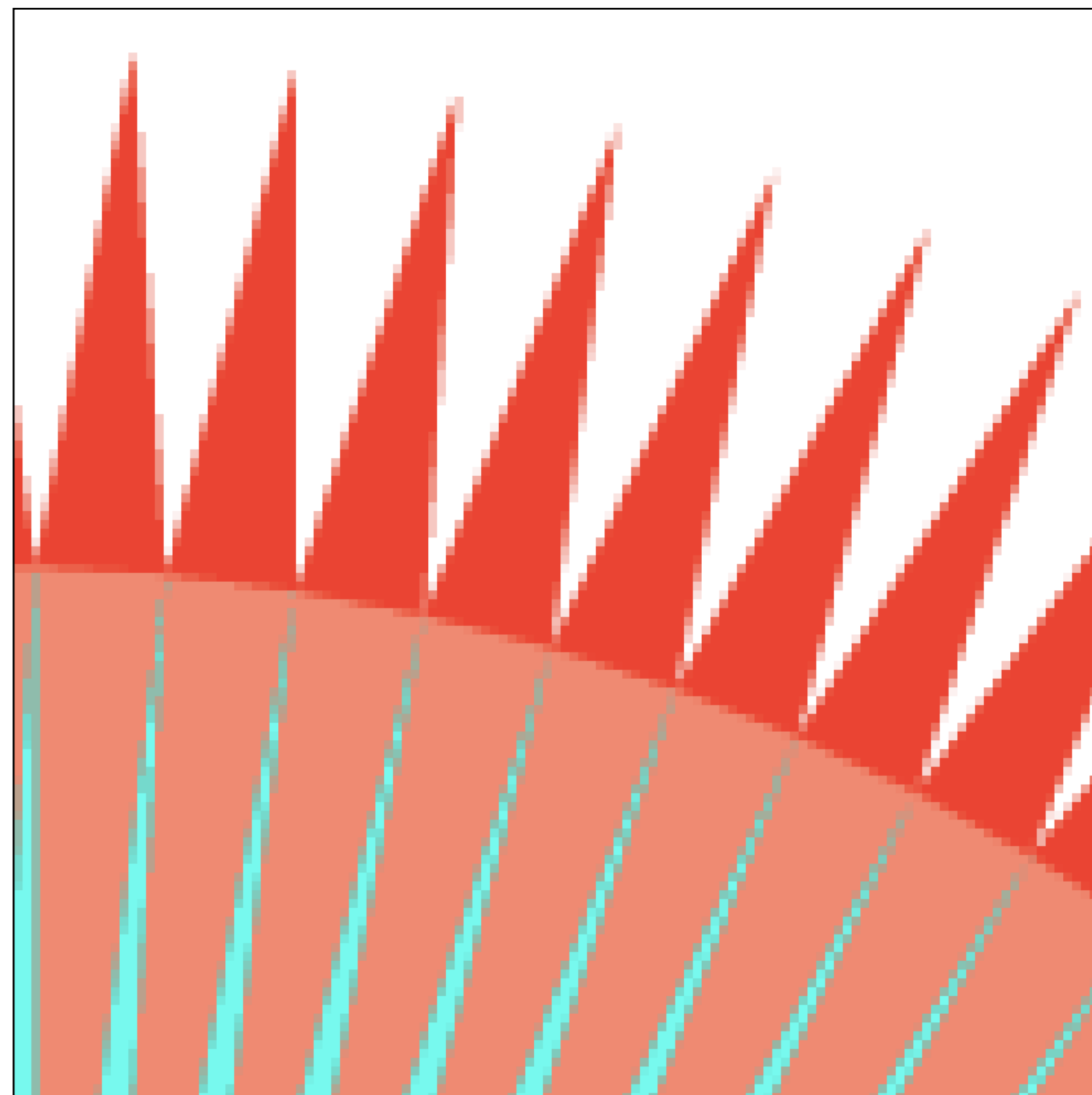
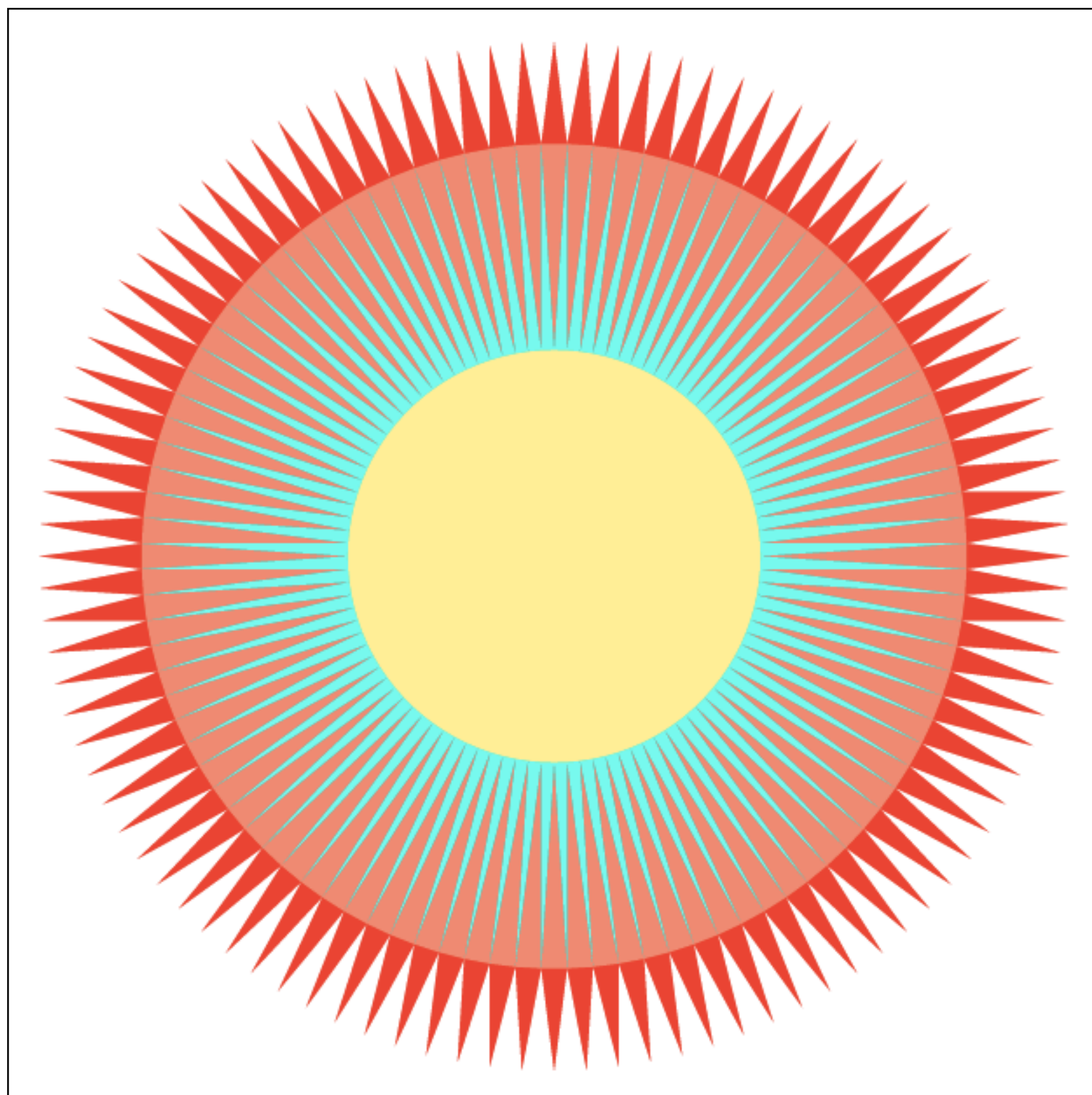
**Note anti-aliased edges of rasterized triangle:  
where pixel values take intermediate values**

# Point sampling

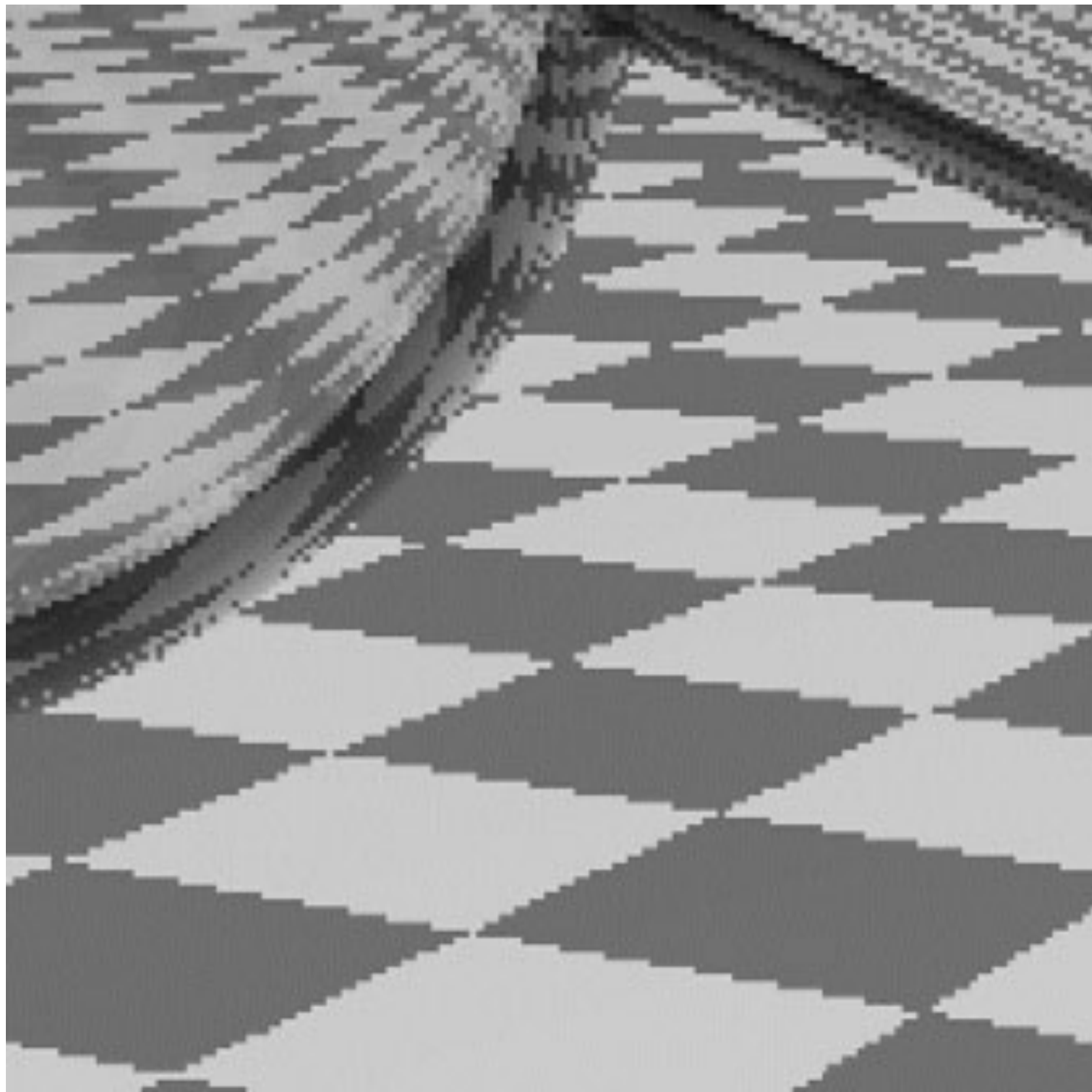


**One sample per pixel**

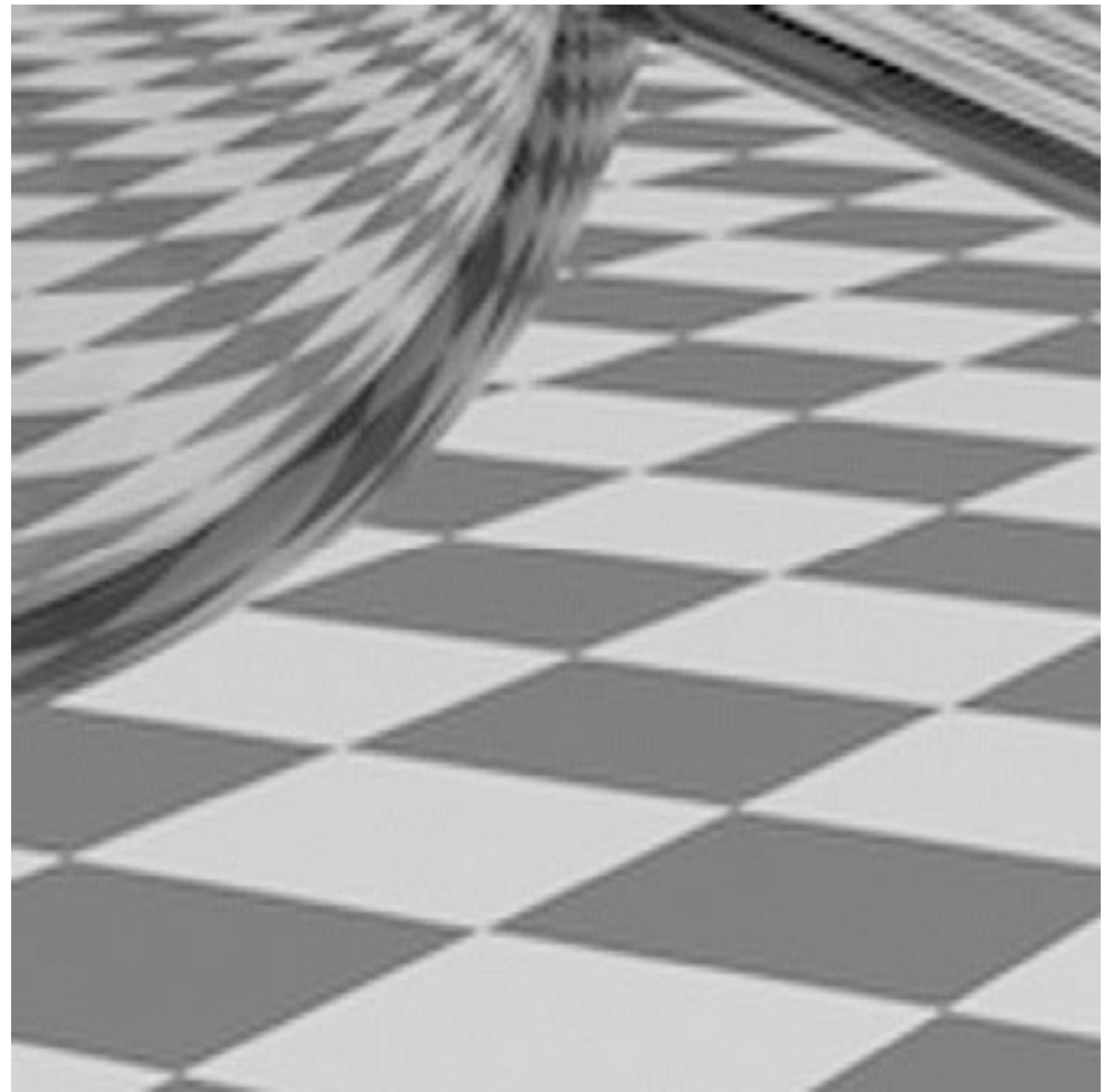
# Anti-aliasing



# Point sampling vs anti-aliasing



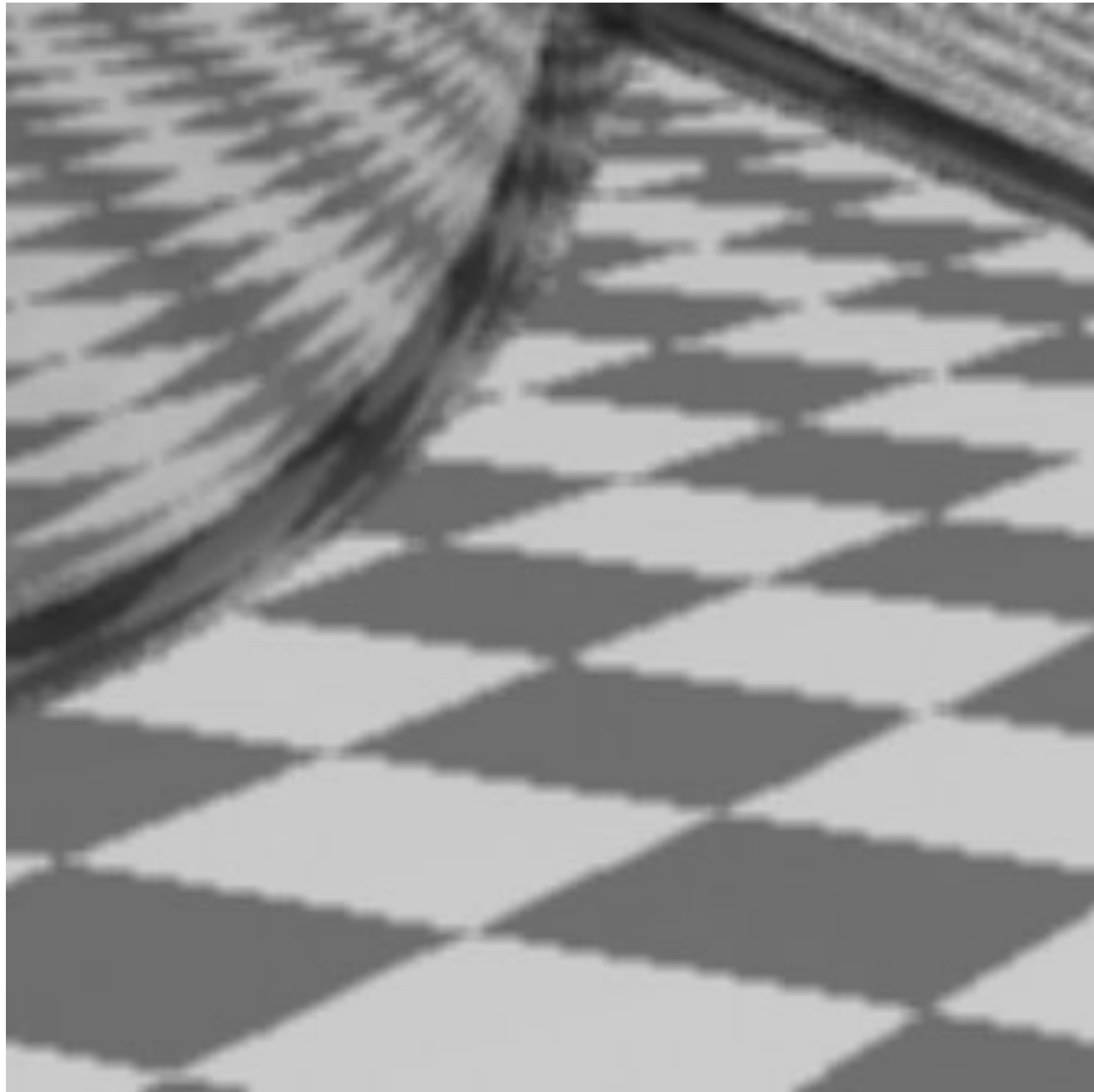
**Jaggies**



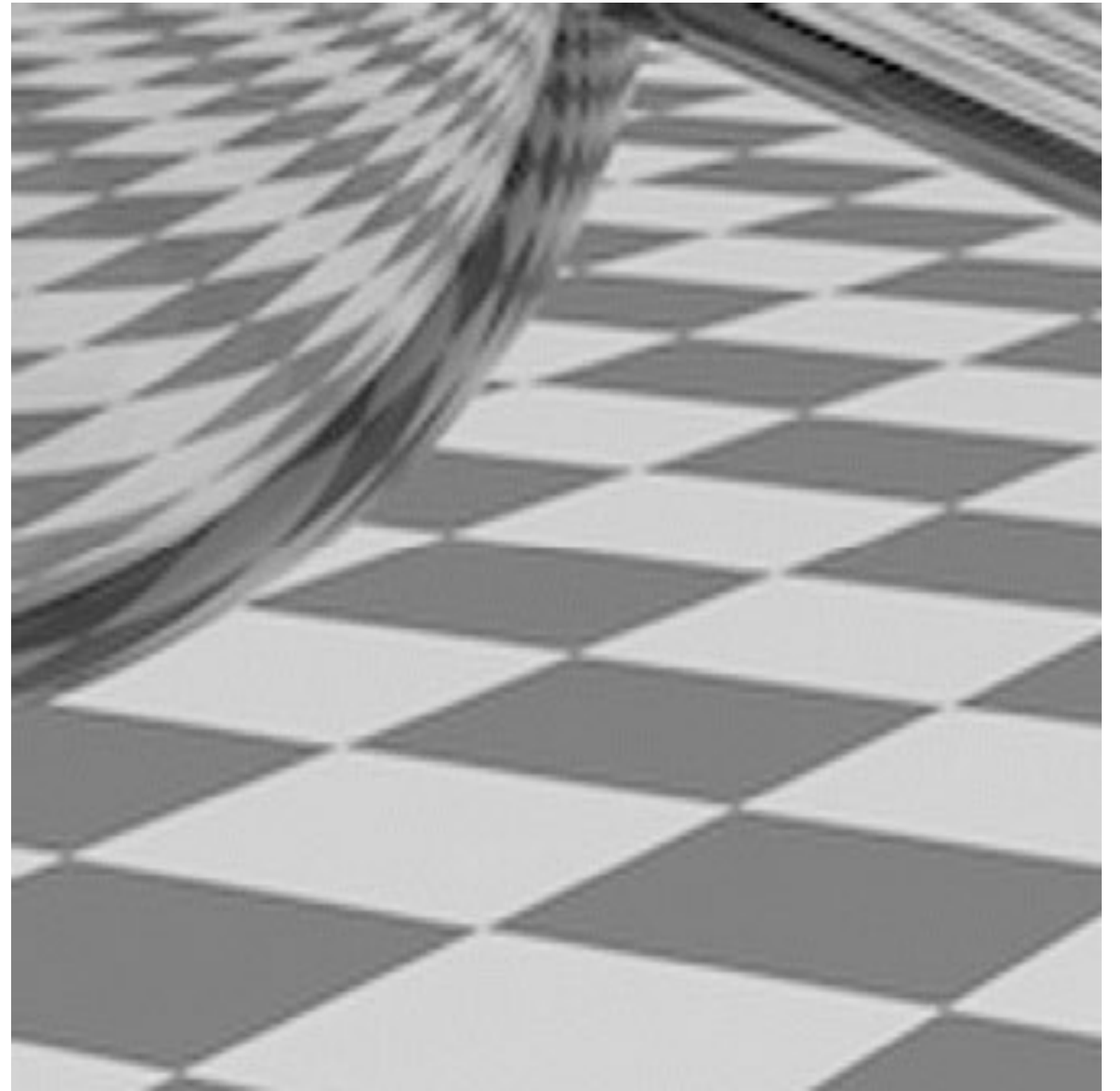
**Pre-filtered**



# Anti-aliasing vs blurring an aliased result



**Blurred Jaggies  
(Sample then filter)**

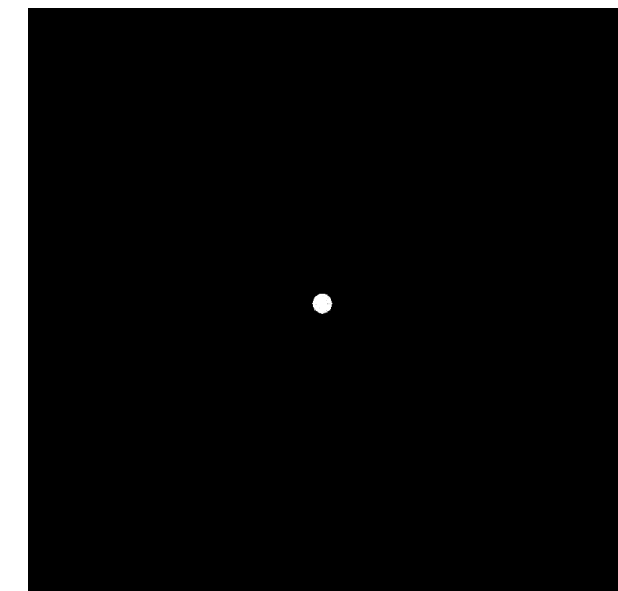
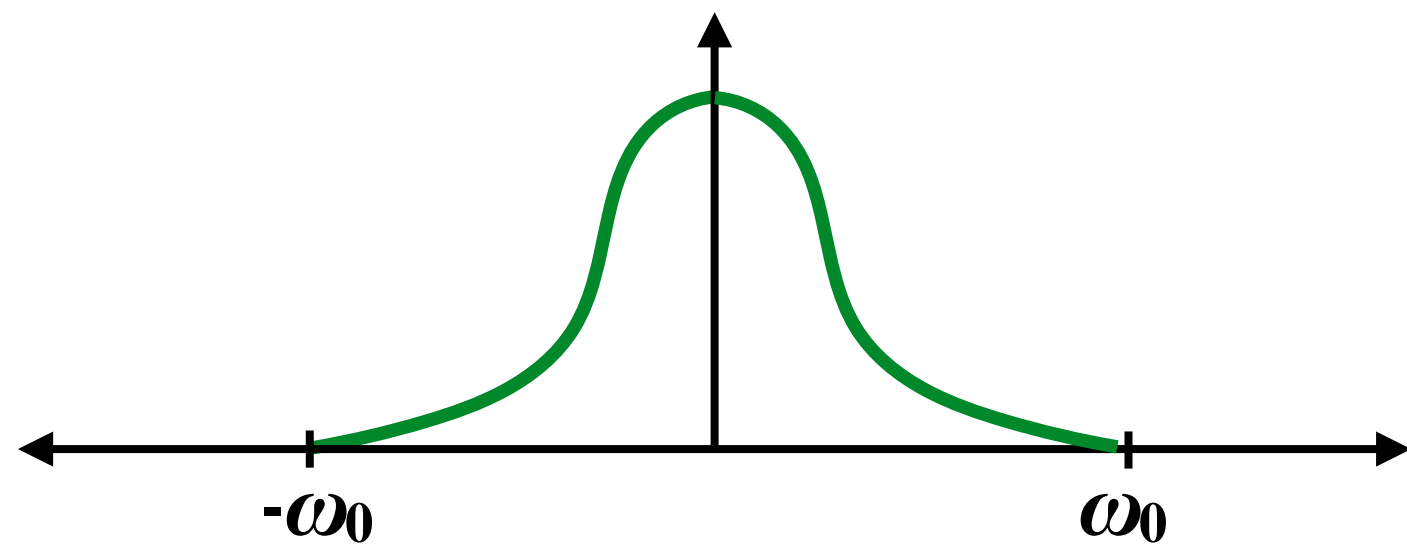


**Pre-Filtered  
(Filter then sample)**

**How much pre-filtering do we need to  
avoid aliasing?**

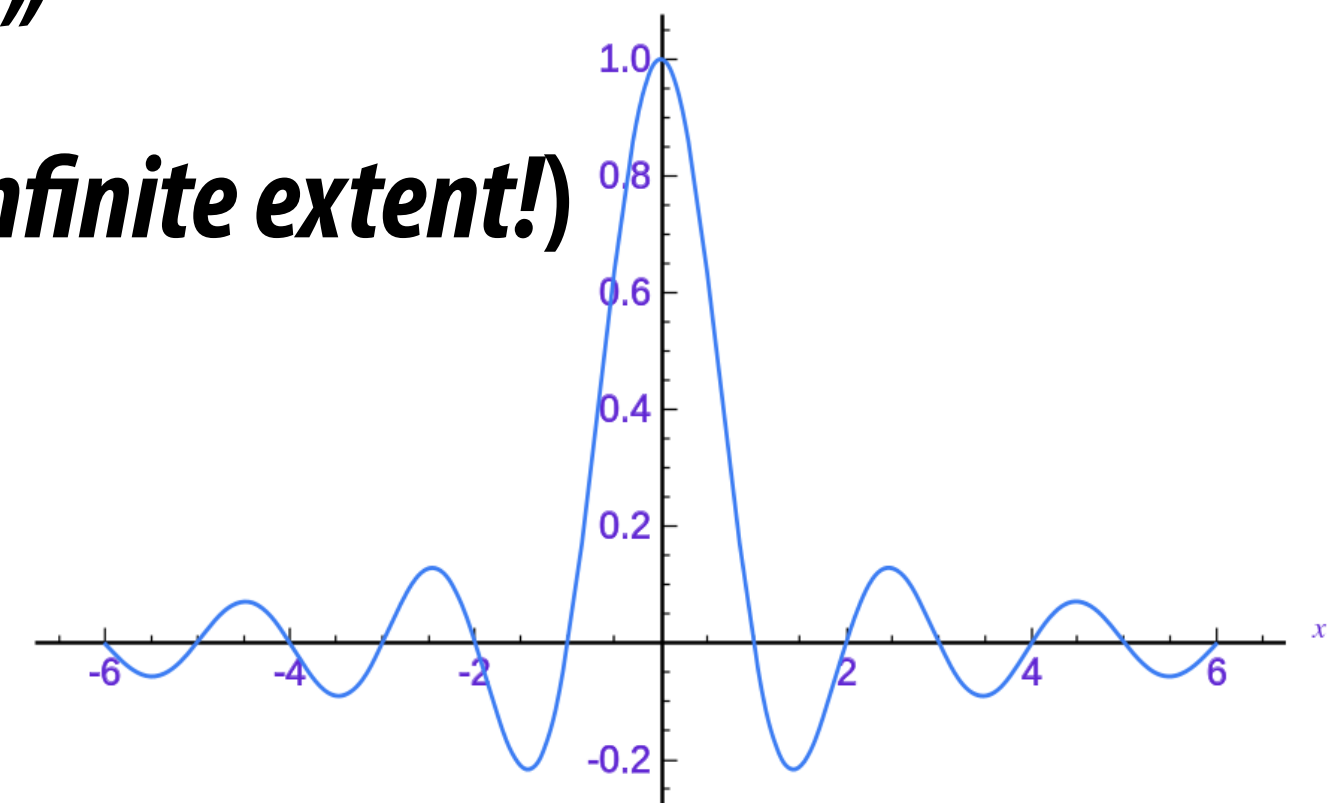
# Nyquist-Shannon theorem

- Consider a band-limited signal: has no frequencies above  $\omega_0$ 
  - 1D: consider low-pass filtered audio signal
  - 2D: recall the blurred image example from a few slides ago



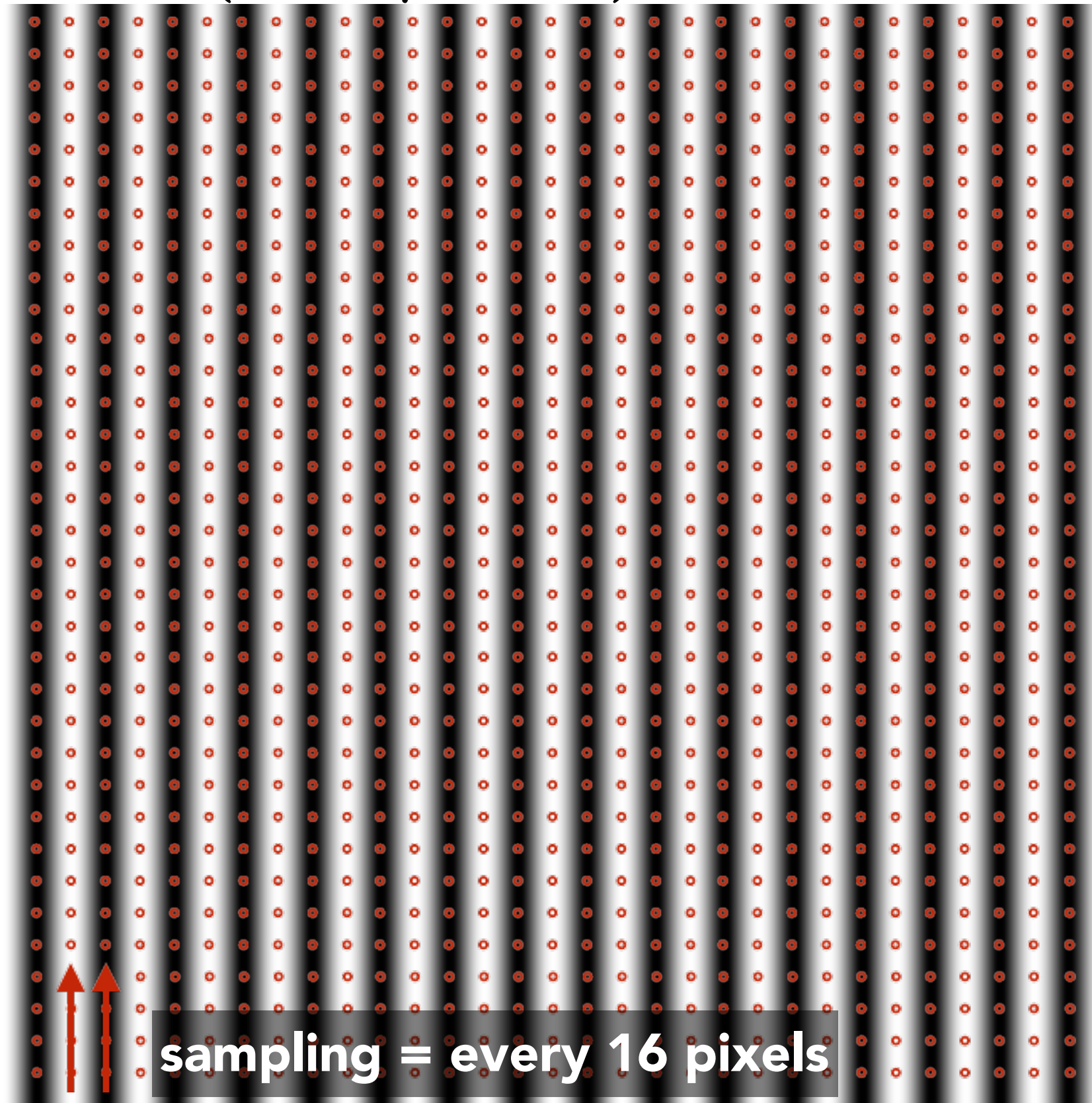
- The signal can be perfectly reconstructed if sampled with period  $T = 1 / 2\omega_0$
- And reconstruction is performed using a “*sinc filter*”
  - Ideal filter with no frequencies above cutoff (*infinite extent!*)

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

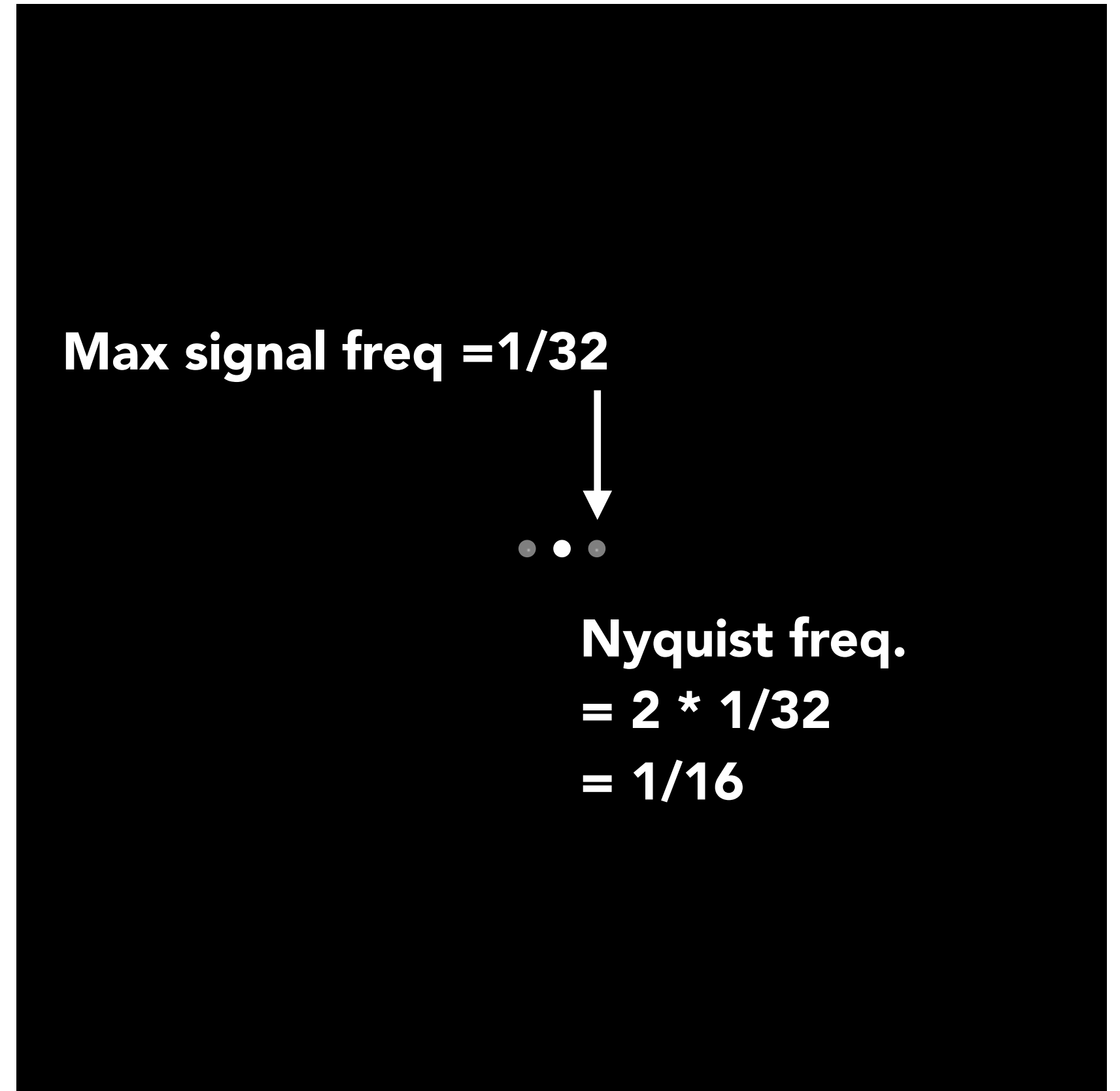


# Signal vs Nyquist frequency: example

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



**Spatial domain**

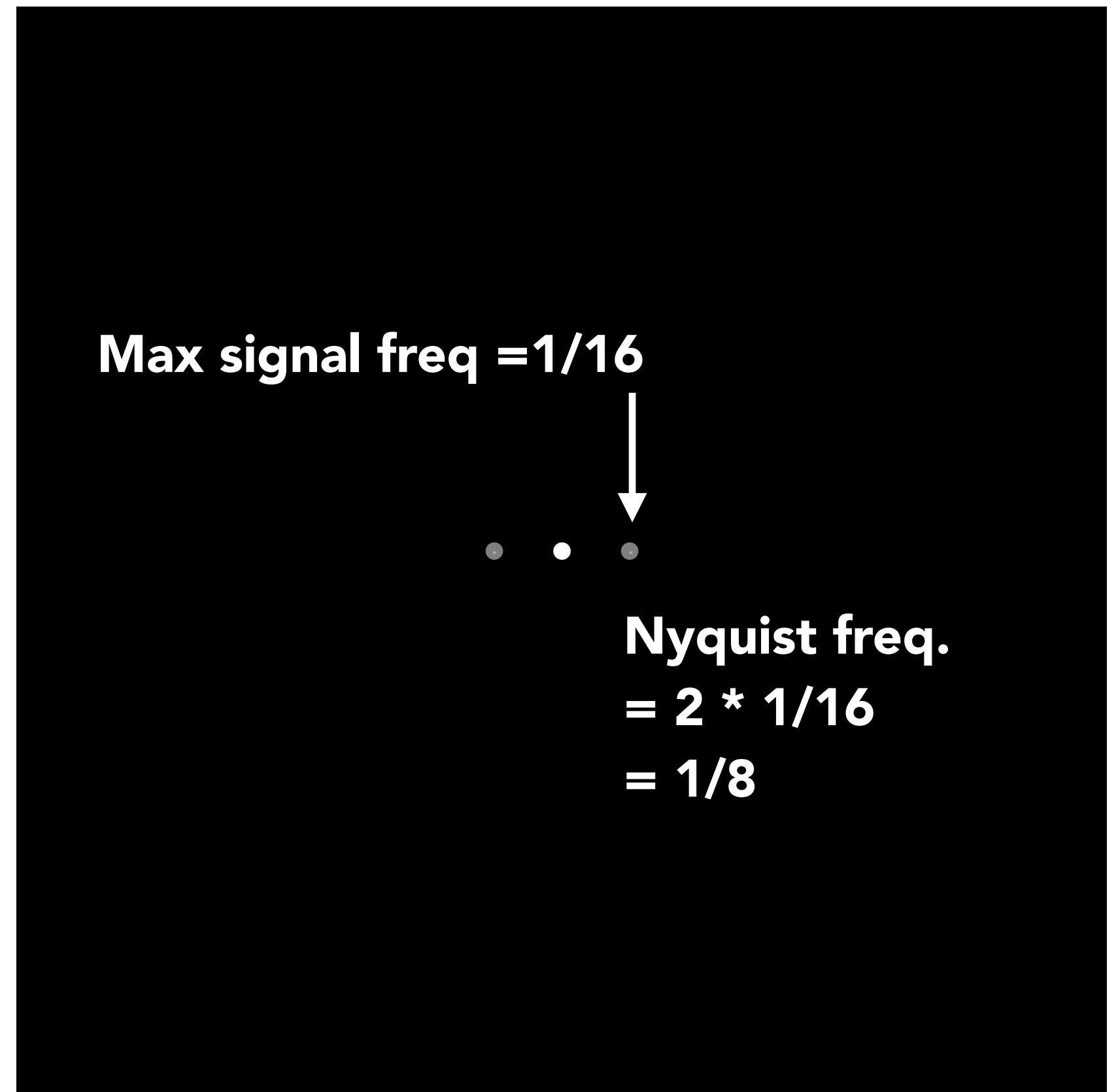
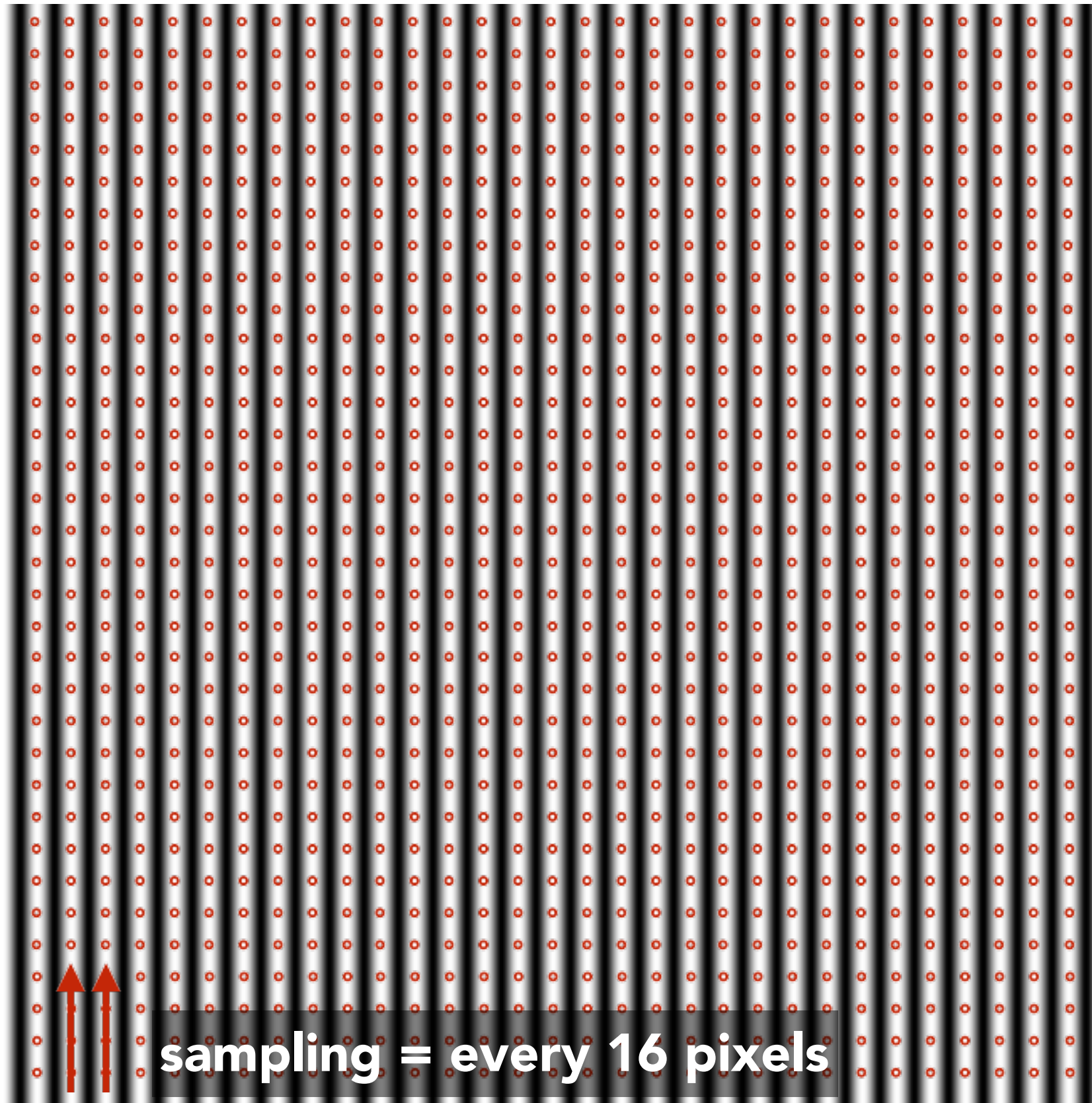


**Frequency domain**

**No Aliasing!**

# Signal vs Nyquist frequency: example

$\sin(2\pi/16)x$  — frequency  $1/16$ ; 16 pixels per cycle



**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)**

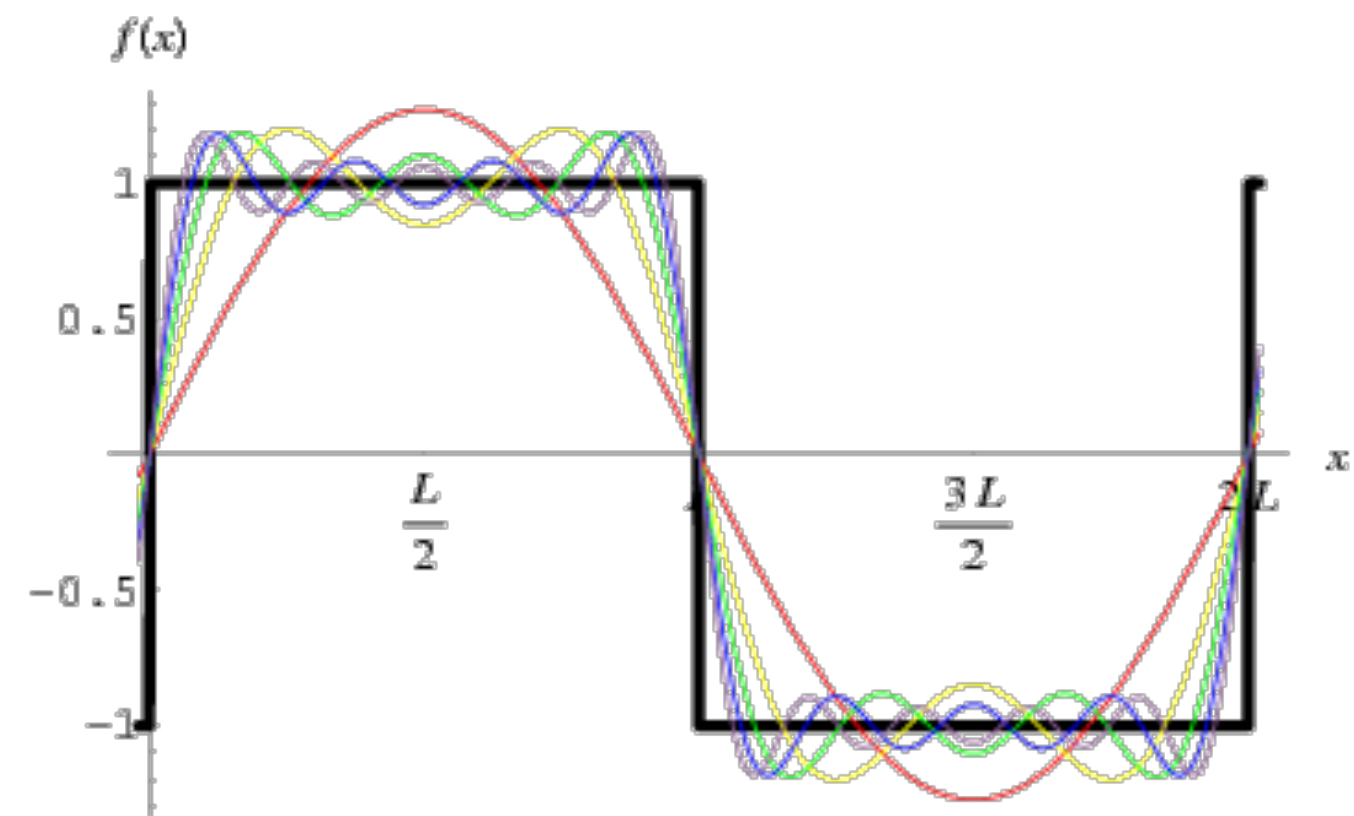
**Consequence: sampling at twice the highest frequency in the signal will eliminate aliasing**

# Challenges of sampling-based approaches in graphics

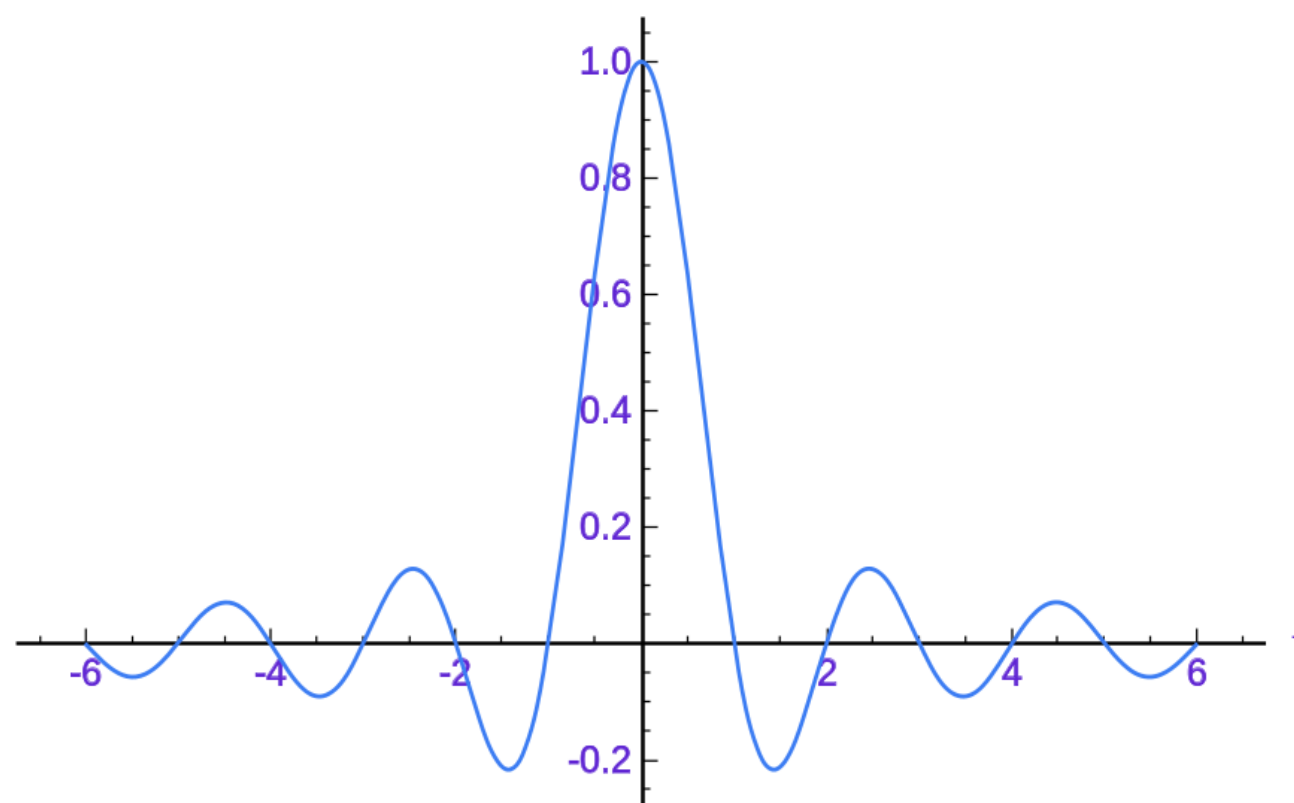
- Our signals are not always band-limited in computer graphics.

Why?

Hint:



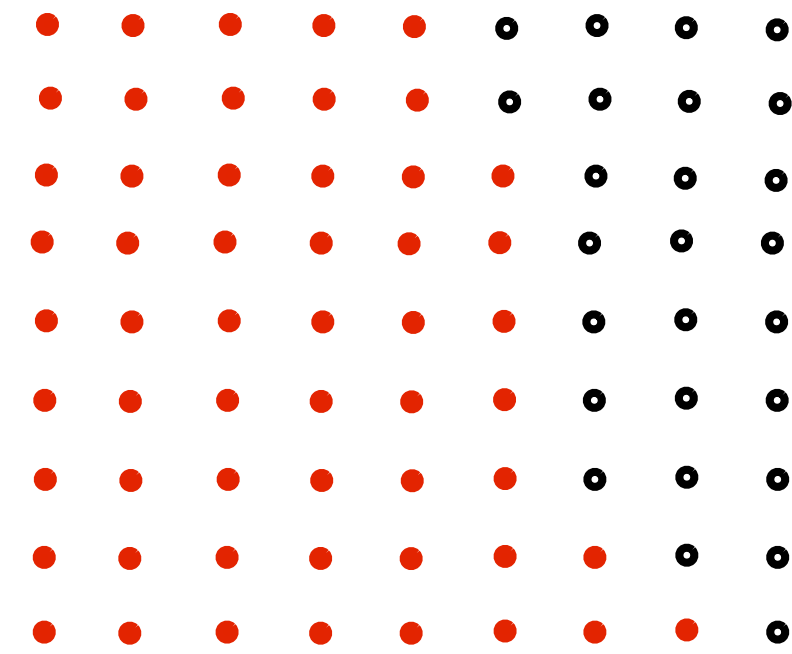
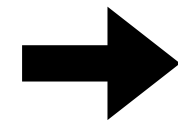
- Also, infinite extent of “ideal” reconstruction filter (sinc) is impractical for efficient implementations. Why?



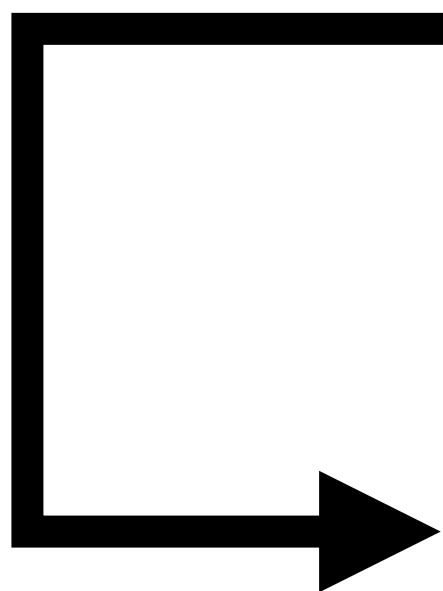
# Recall our anti-aliasing technique in the first half of lecture



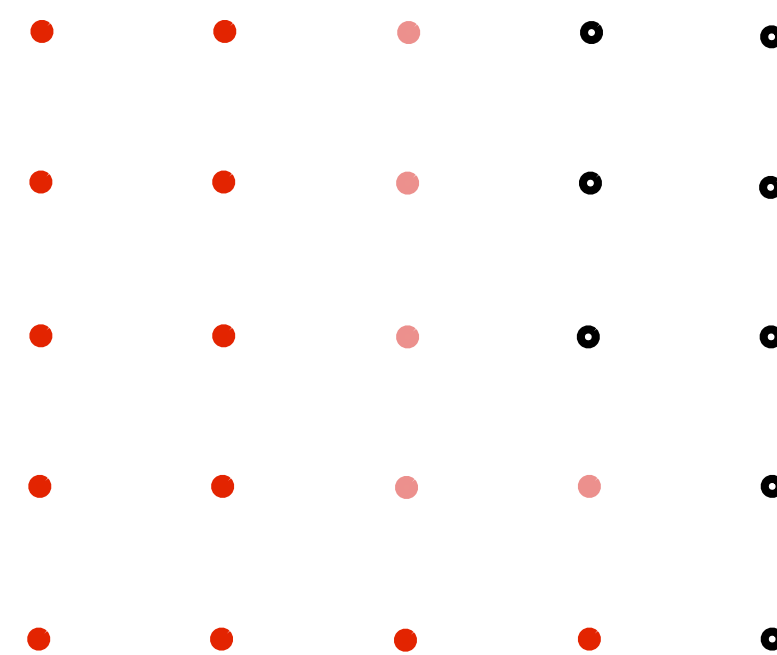
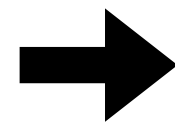
Original signal  
(high frequency edge)



Dense sampling of signal



Reconstructed signal  
(after averaging over pixel)



Coarsely sampled signal



**Filtering = convolution**

# Convolution

Signal

1	3	5	3	7	1	3	8	6	4
---	---	---	---	---	---	---	---	---	---

Filter

1	2	1
---	---	---

# Convolution

Signal

1	3	5	3	7	1	3	8	6	4
---	---	---	---	---	---	---	---	---	---

Filter

1	2	1
---	---	---

$$1 \times 1 + 3 \times 2 + 5 \times 1 = 12$$

Result

12									
----	--	--	--	--	--	--	--	--	--

# Convolution

Signal

1	3	5	3	7	1	3	8	6	4
---	---	---	---	---	---	---	---	---	---

Filter

1	2	1
---	---	---

$$3 \times 1 + 5 \times 2 + 3 \times 1 = 16$$

Result

12	16								
----	----	--	--	--	--	--	--	--	--

# Convolution

Signal

1	3	5	3	7	1	3	8	6	4
---	---	---	---	---	---	---	---	---	---

Filter

1	2	1
---	---	---

$$5 \times 1 + 3 \times 2 + 7 \times 1 = 18$$

Result

12	16	18							
----	----	----	--	--	--	--	--	--	--

# Discrete 2D convolution

$$(f * g)(x, y) = \sum_{i, j = -\infty}^{\infty} f(i, j) I(x - i, y - j)$$

output image                      filter                      input image

Consider  $f(i, j)$  that is nonzero only when:  $-1 \leq i, j \leq 1$

Then:

$$(f * g)(x, y) = \sum_{i, j = -1}^1 f(i, j) I(x - i, y - j)$$

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

$$f(i, j) = \mathbf{F}_{i, j} \quad \text{(often called: "filter weights", "filter kernel")}$$

# Box filter (used in a 2D convolution)

$$\frac{1}{9}$$

1	1	1
1	1	1
1	1	1

**Example: 3x3 box filter**

# 2D convolution with box filter blurs the image



**Original image**



**Blurred  
(convolve with box filter)**

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



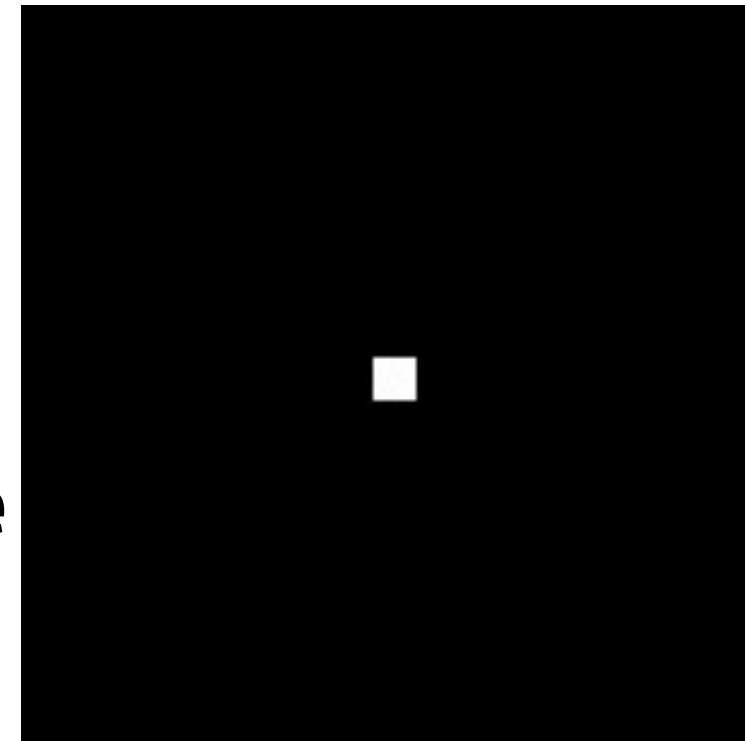
# Convolution theorem

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

Spatial  
Domain



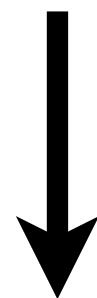
$*$   
convolve



=



Fourier  
Transform



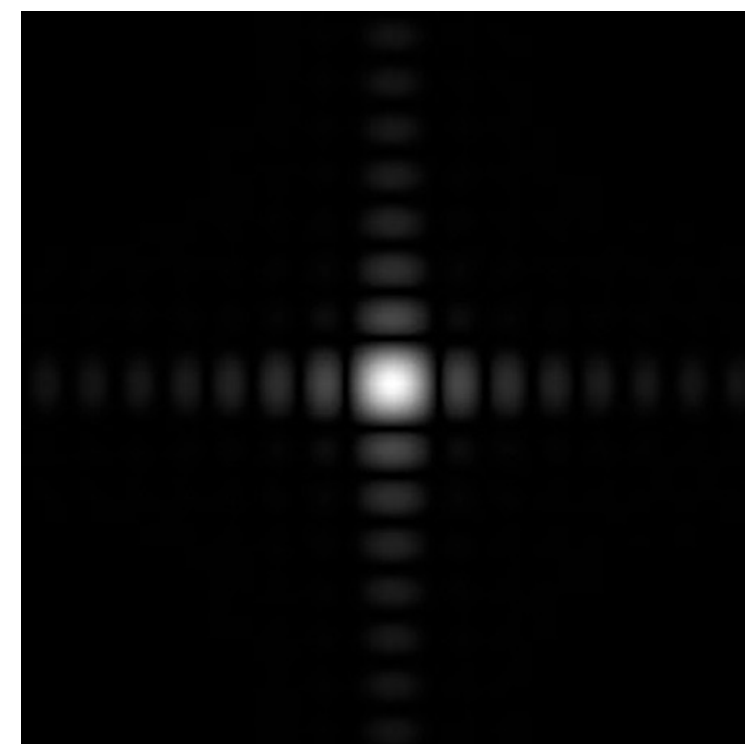
Inv. Fourier  
Transform



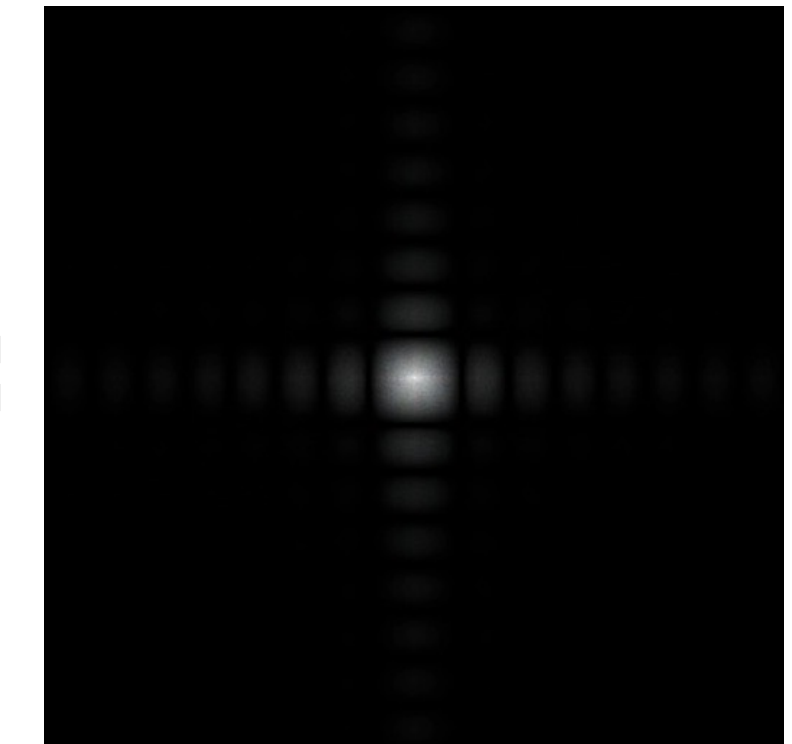
Frequency  
Domain



$\times$



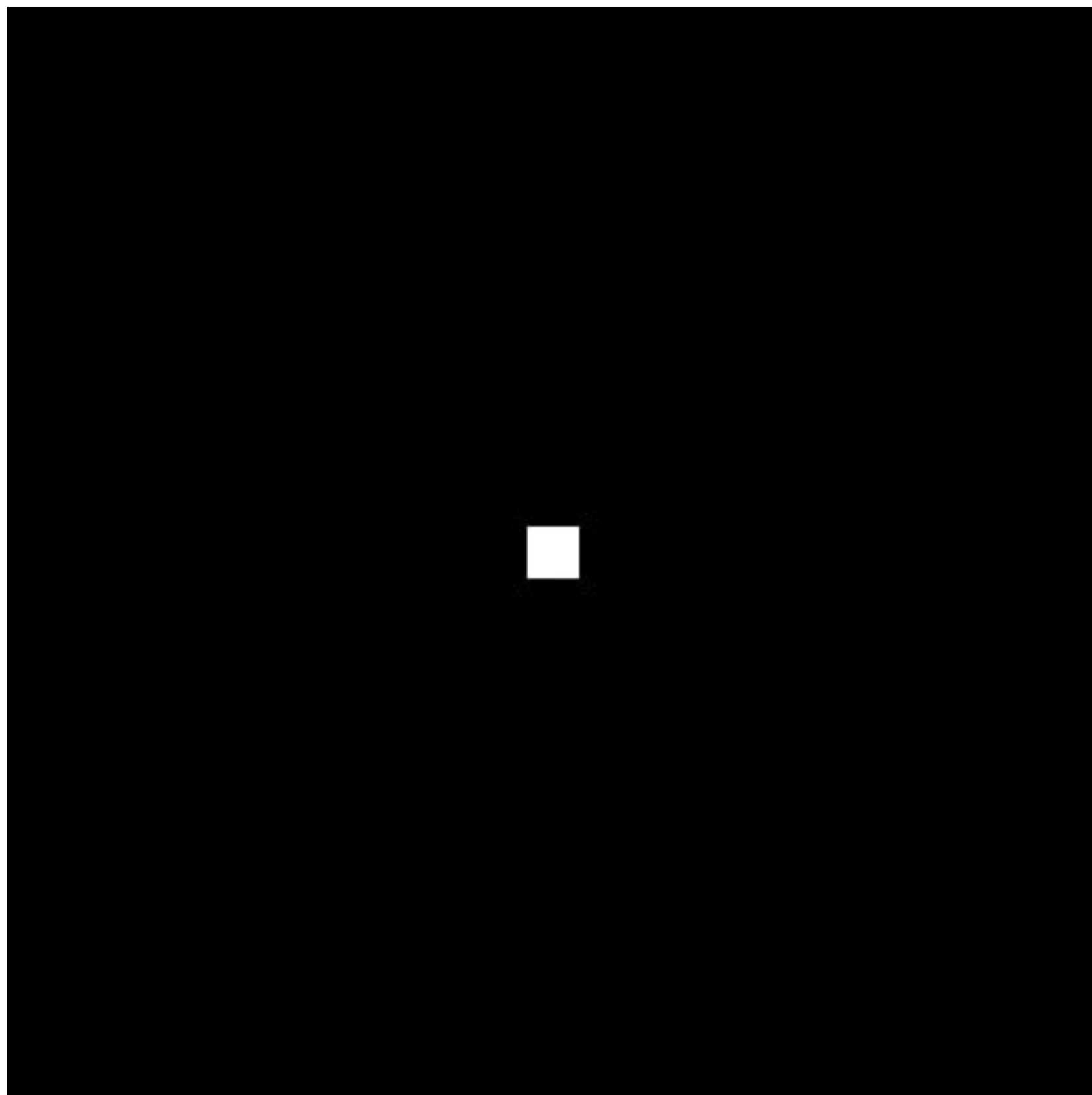
=



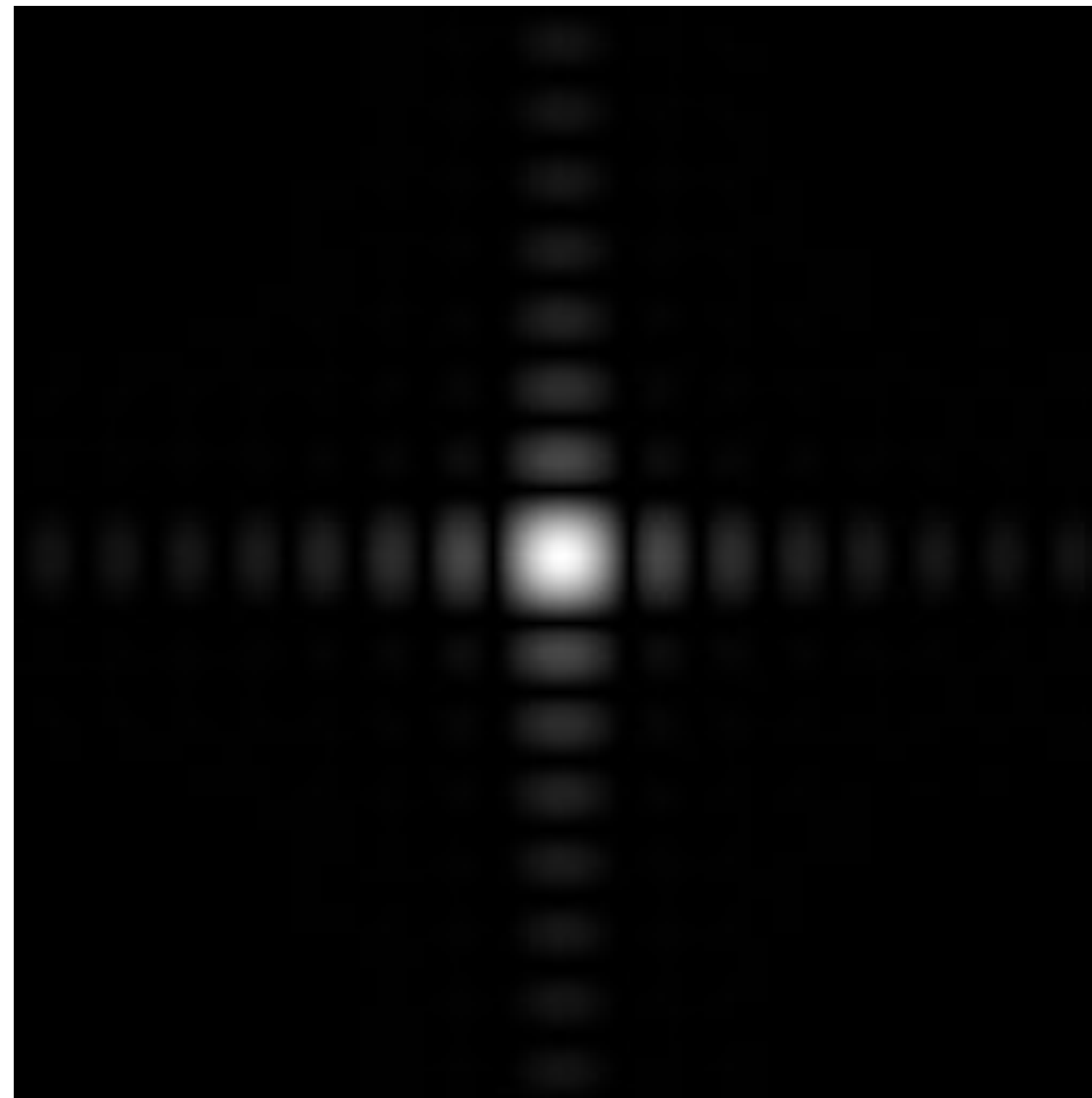
# Convolution theorem

- **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

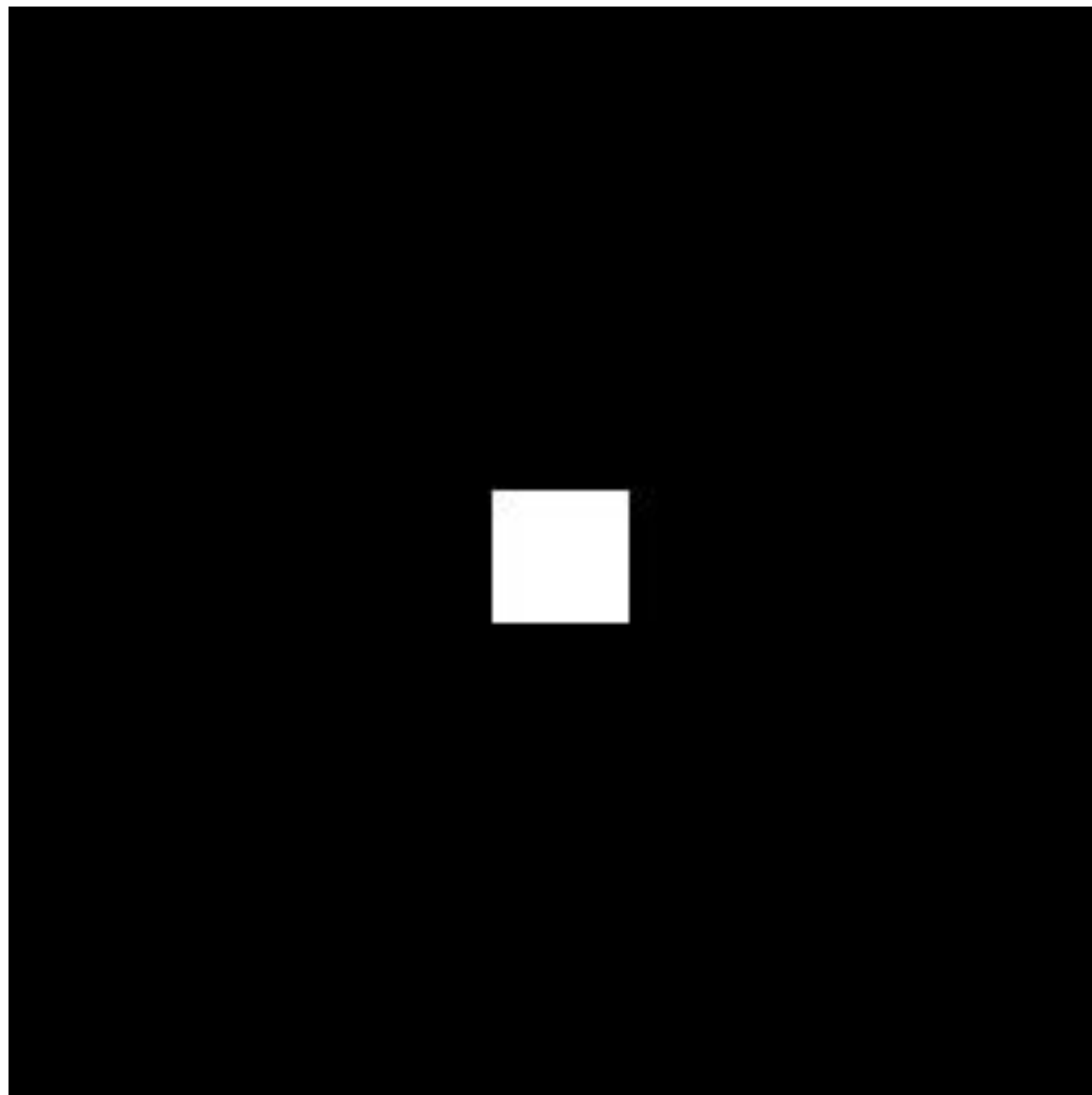


**Spatial domain**

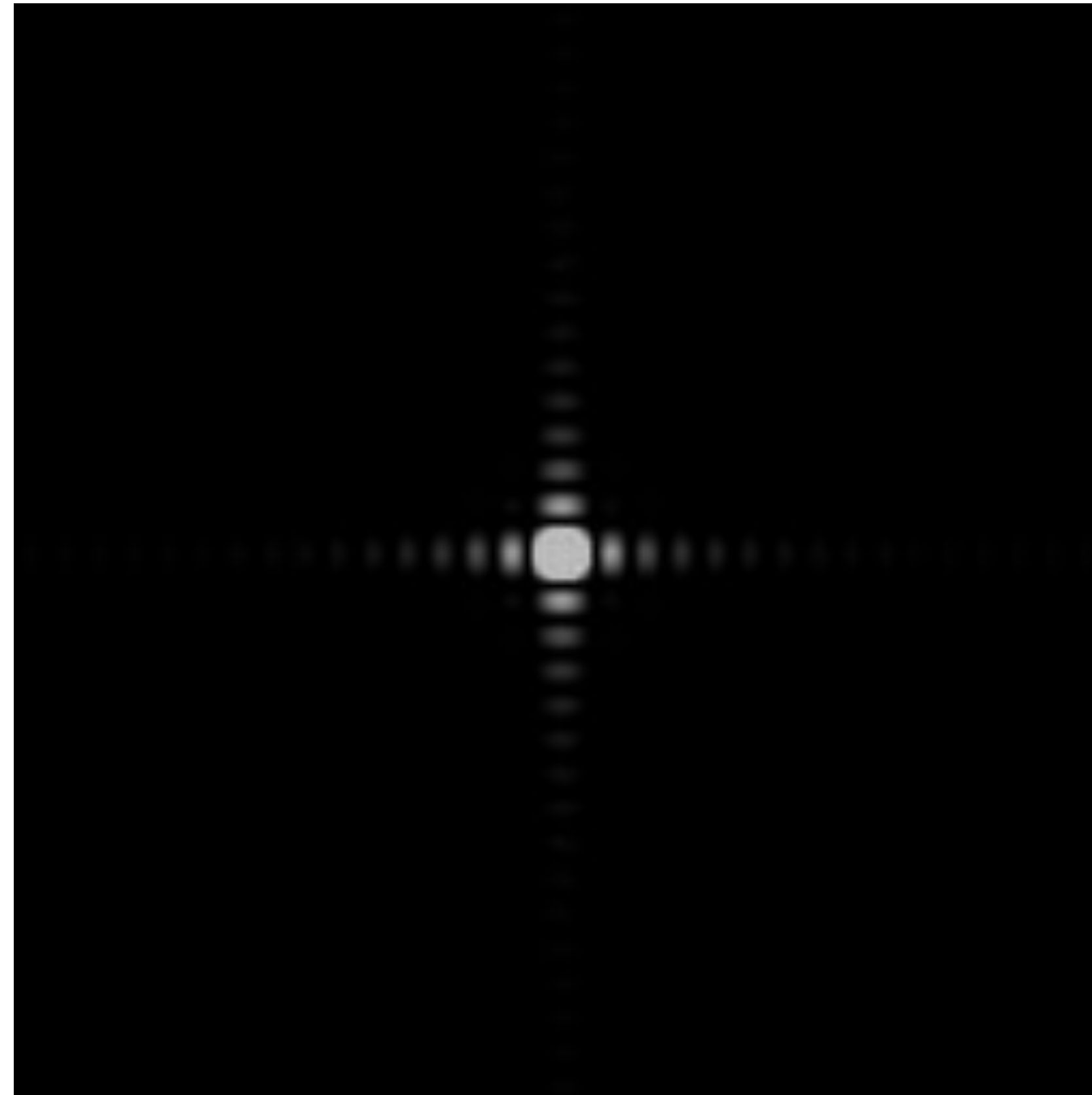


**Frequency domain**

# Wider filter kernel = 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 (increase Nyquist frequency)**
  - **Higher resolution displays, sensors, framebuffers...**
  - **But: costly and may need very high resolution**
  
- **Anti-aliasing**
  - **Simple idea: remove (or reduce) signal frequencies above the Nyquist frequency 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 average value inside a pixel area of  $f(x,y) = \text{inside}(\text{tri},x,y)$  is equal to the area of the pixel covered by the triangle.

**Original**



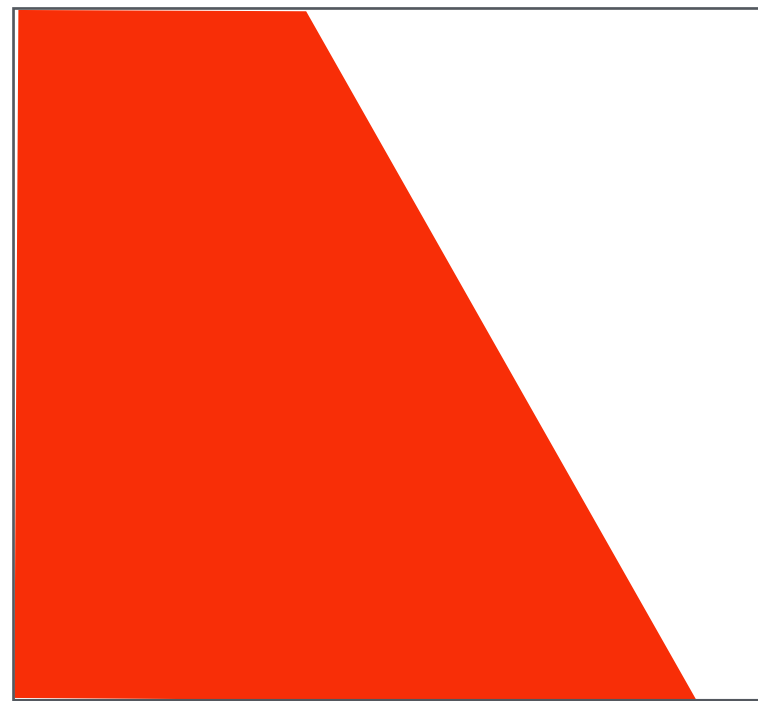
**Filtered**



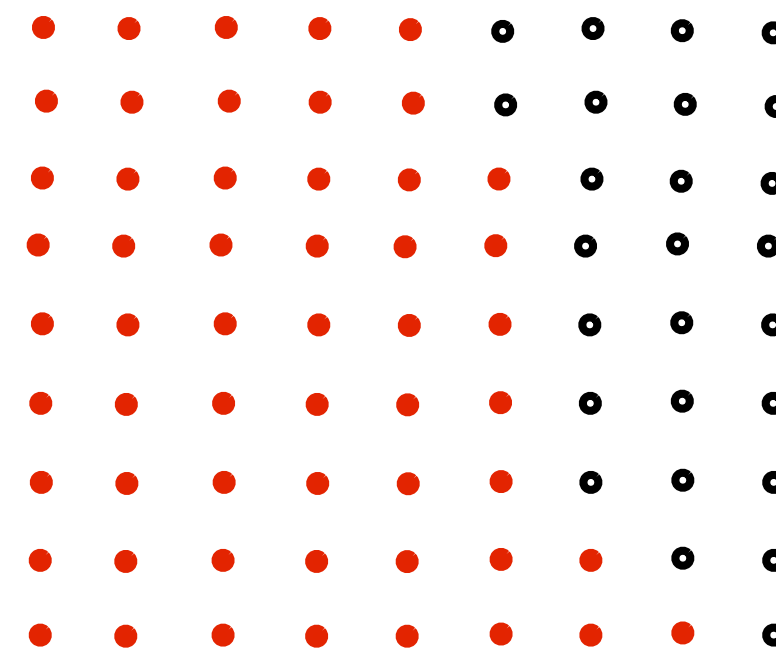
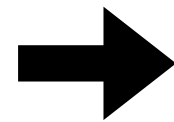
←→  
1 pixel width



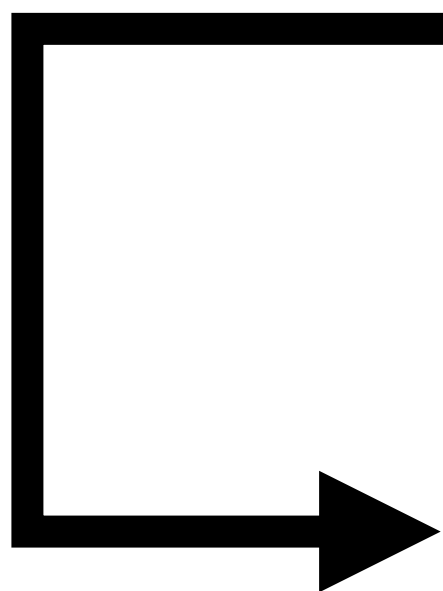
# Putting it all together: anti-aliasing via supersampling



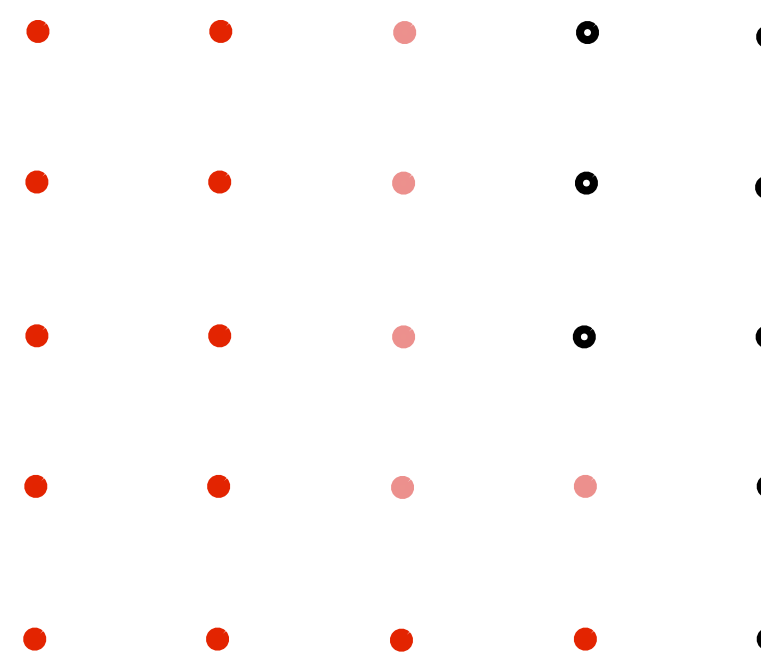
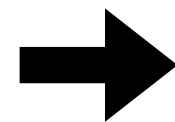
Original signal  
(with high frequency edge)



Dense sampling of signal  
(supersampling)



Reconstructed signal  
(averaging over pixel (via convolution) yields  
new signal with high frequencies removed)



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...**

# Acknowledgements

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