

# ECE 4973/5973: Lecture 6

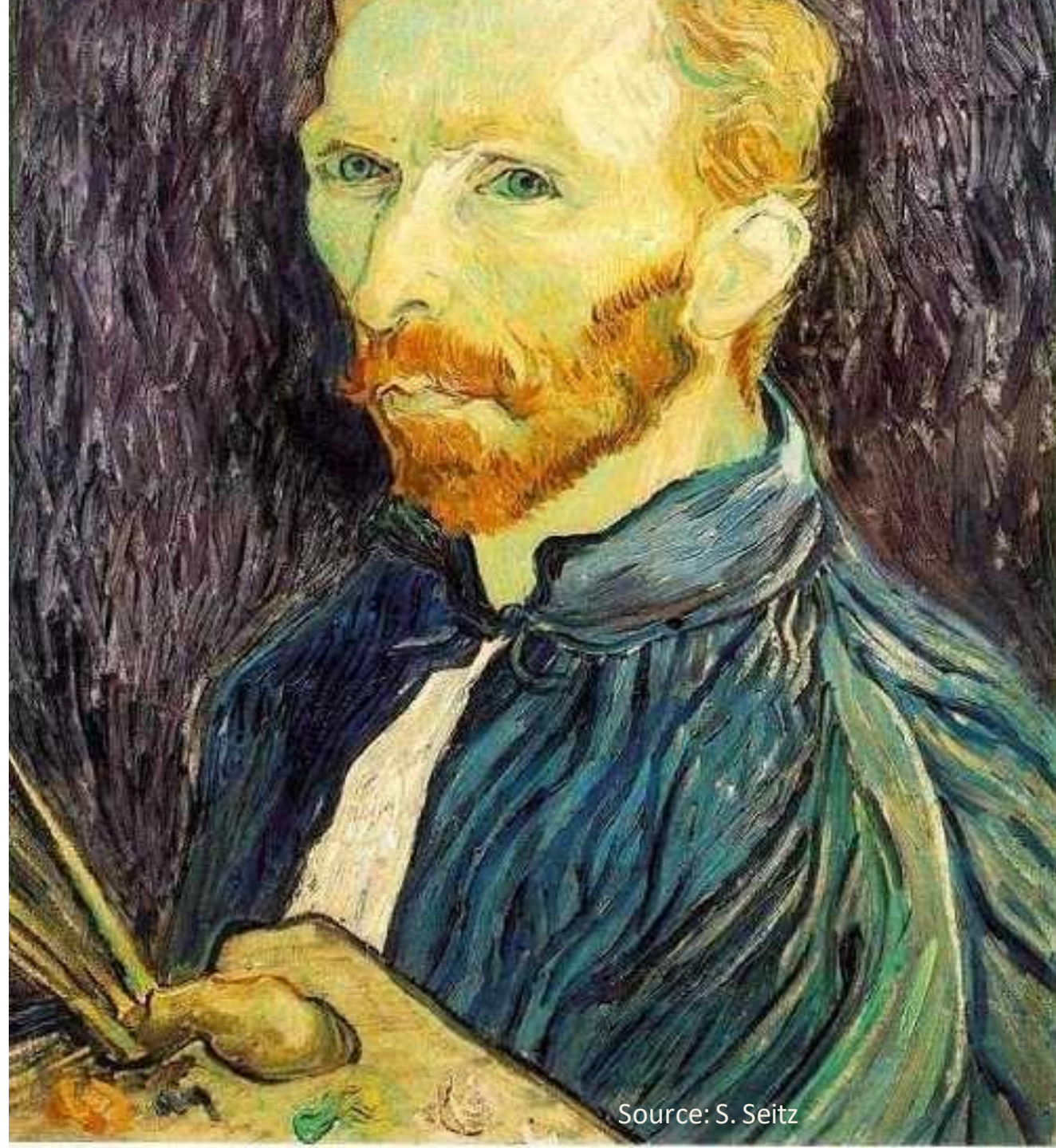
## Resampling

Samuel Cheng

Slide credits: Noah Snavely

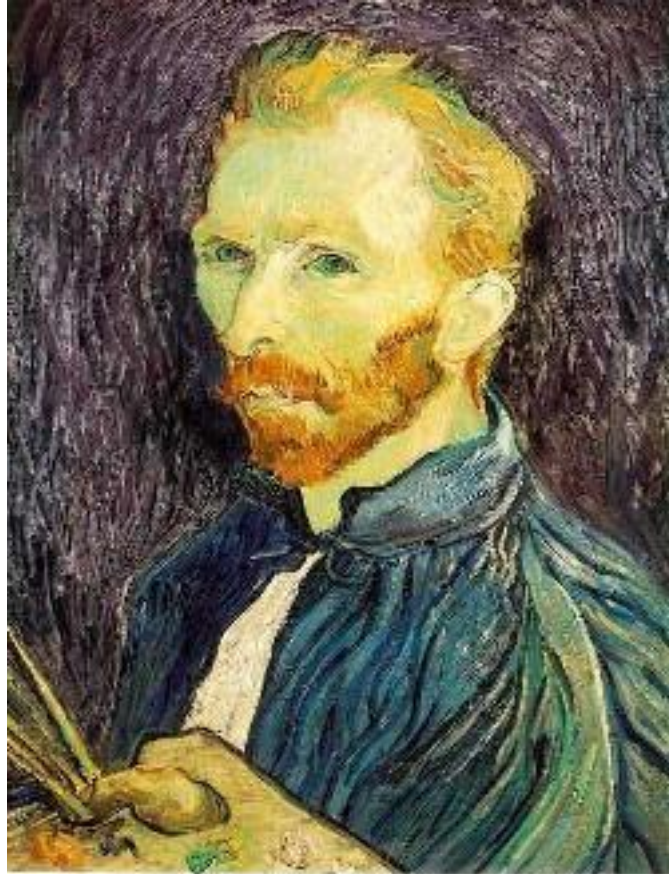
# Image Scaling

This image is too big to fit on the screen. How can we generate a half-sized version?



Source: S. Seitz

# Image sub-sampling



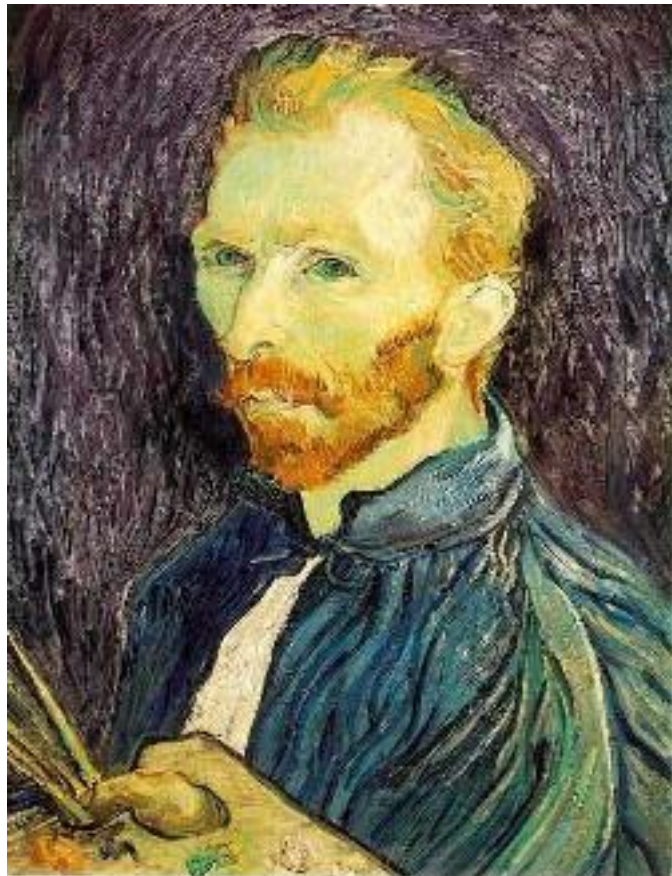
1/4



1/8

Throw away every other row and column to create a 1/2 size image  
- called *image sub-sampling*

# Image sub-sampling



1/2



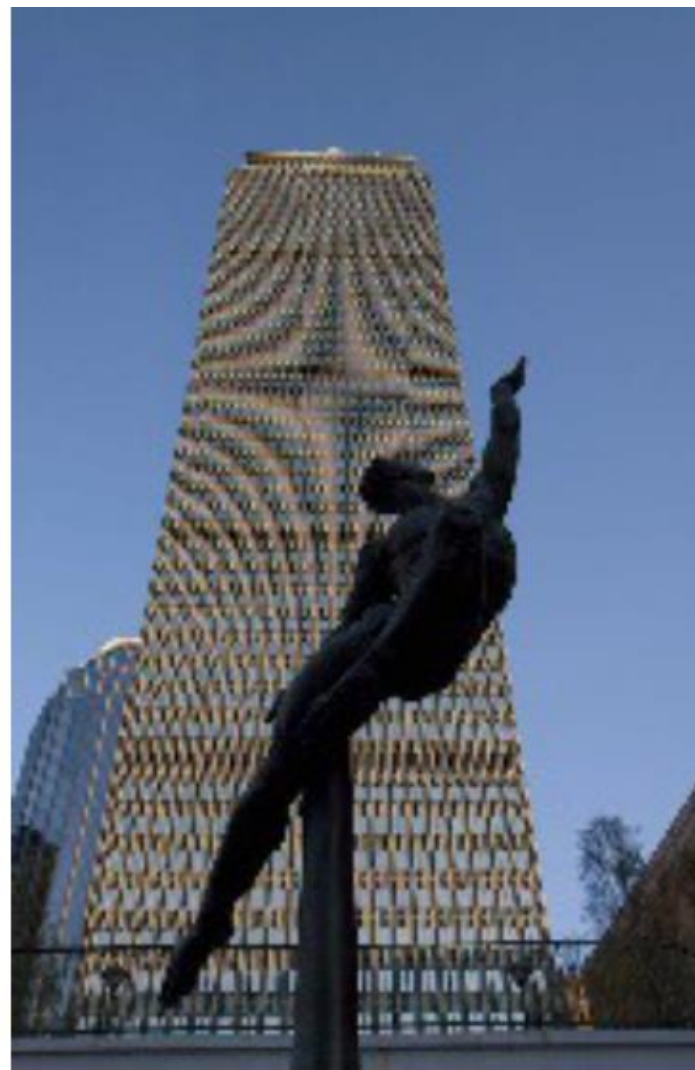
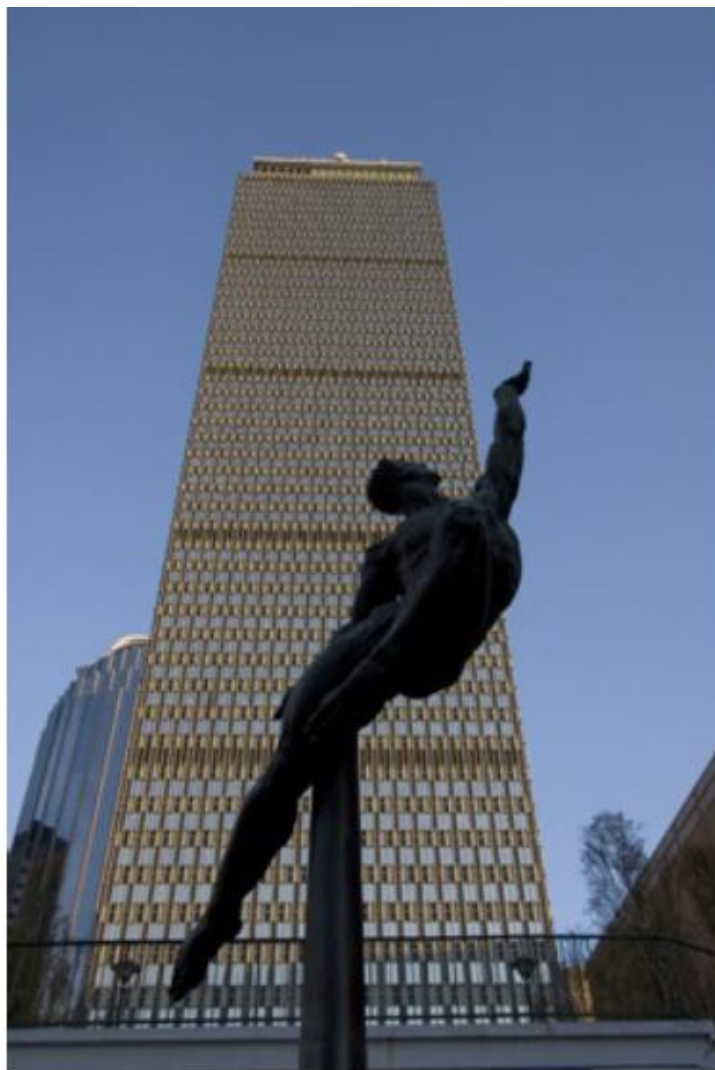
1/4 (2x zoom)



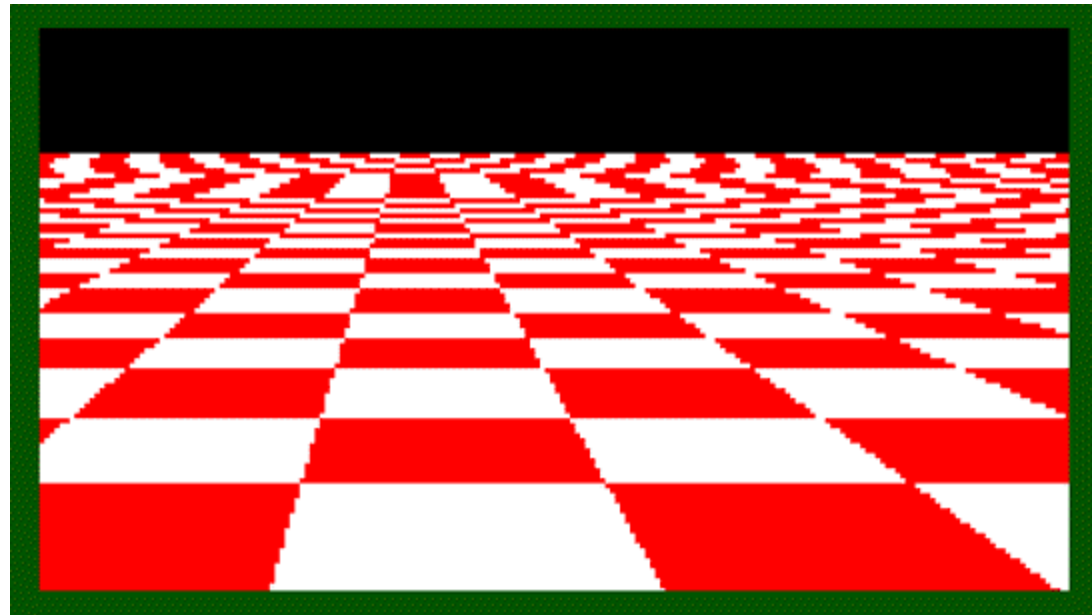
1/8 (4x zoom)

Why does this look so cruffy?

# Image sub-sampling



# Even worse for synthetic images







The blue and green colors are actually the same

<http://blogs.discovermagazine.com/badastronomy/2009/06/24/the-blue-and-the-green/>



# Artifacts from sampling

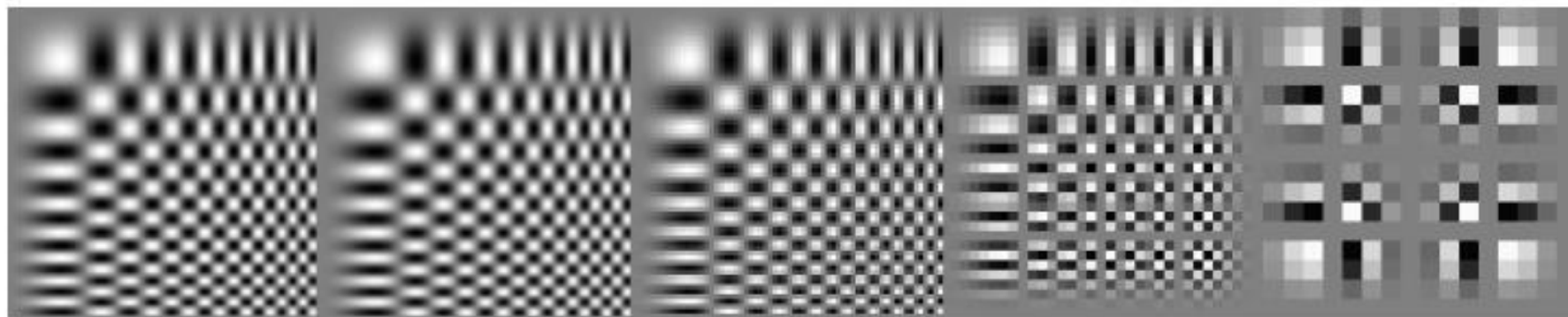
256x256

128x128

64x64

32x32

16x16



## Interesting videos

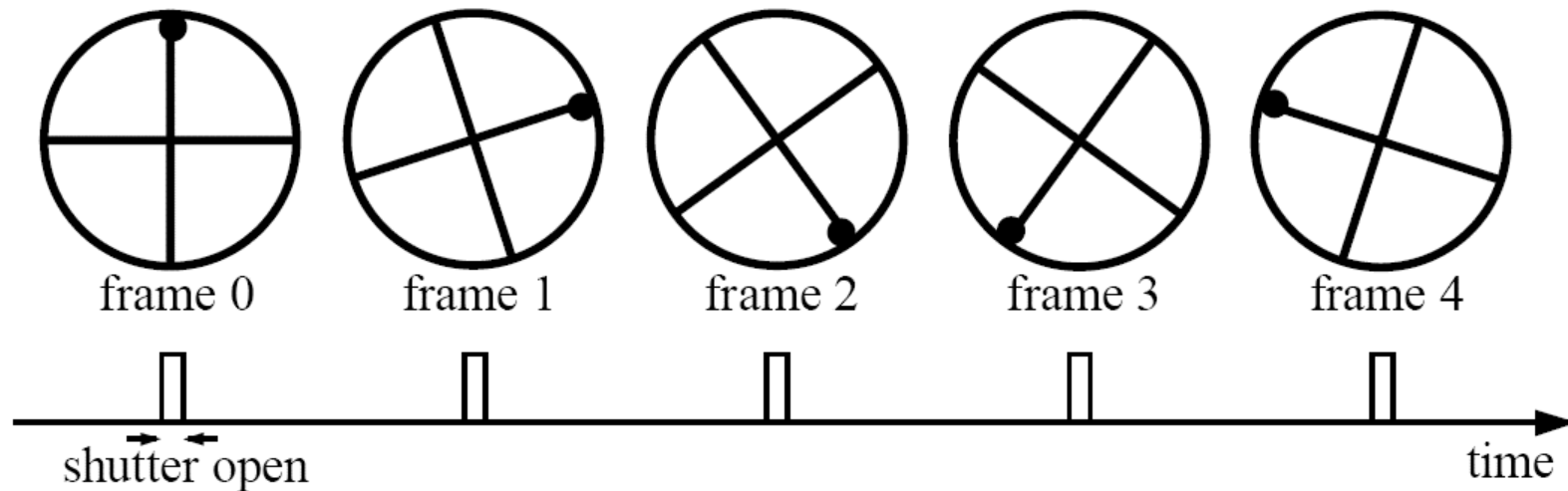


# Aliasing in video

Imagine a spoked wheel moving to the right (rotating clockwise).

Mark wheel with dot so we can see what's happening.

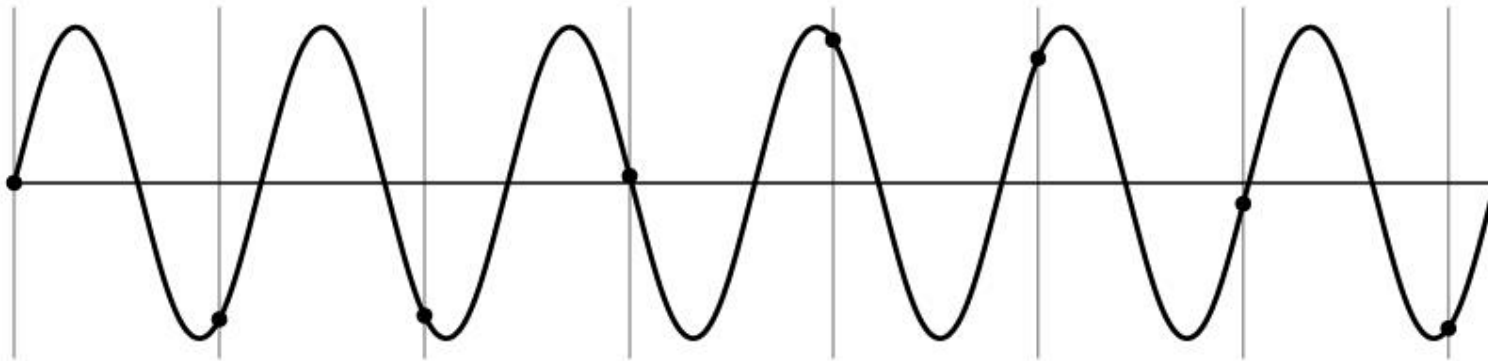
If camera shutter is only open for a fraction of a frame time (frame time = 1/30 sec. for video, 1/24 sec. for film):



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

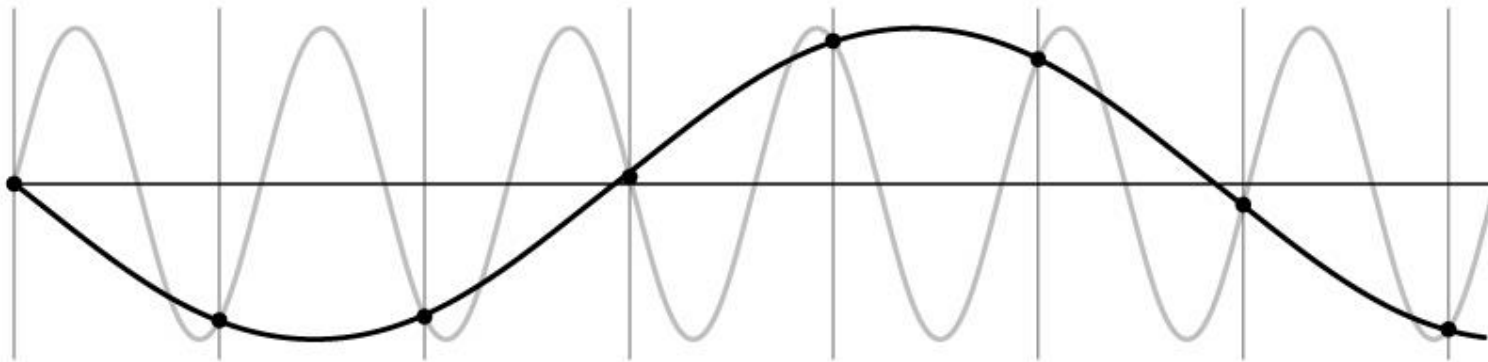
# Aliasing problem

- 1D example (sinewave):



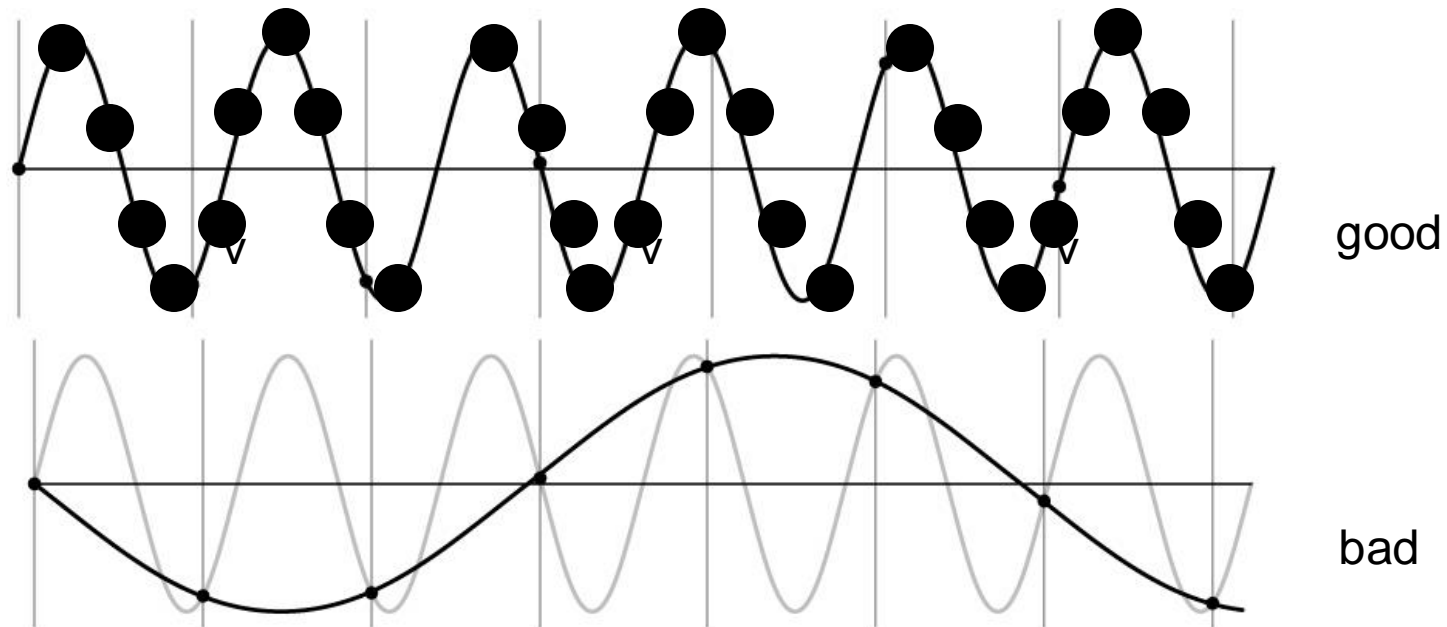
# Aliasing problem

- 1D example (sinewave):

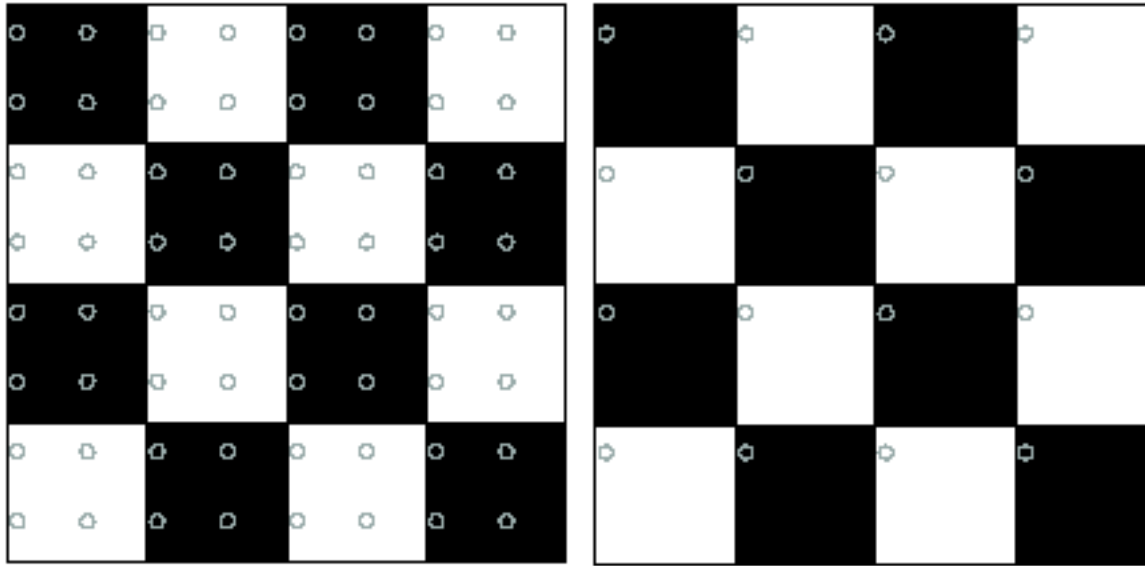


# Nyquist-Shannon Sampling Theorem

- When sampling a signal at discrete intervals, the sampling frequency must be  $\geq 2 \times f_{\max}$
- $f_{\max}$  = max frequency

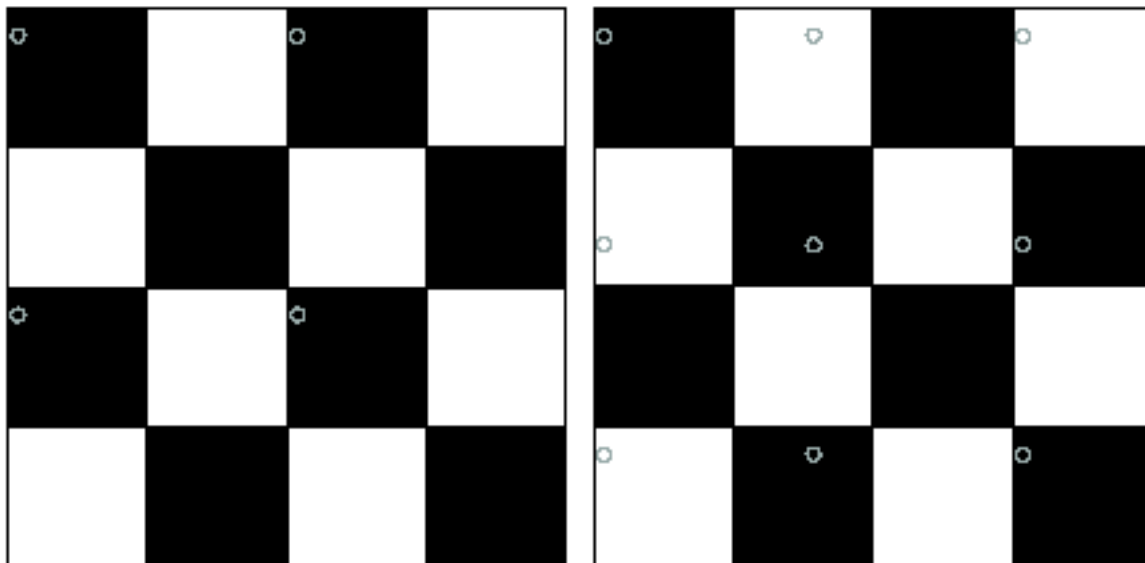


# Nyquist limit – 2D example



Good sampling

How to sample?



Bad sampling

Why aliasing precisely??

# Revisit FT

$$f(t) \xrightarrow{\mathcal{F}} F(\omega) = \int_{-\infty}^{\infty} f(t) e^{-i\omega t} dt$$

$$F(\omega) \xrightarrow{\mathcal{F}^{-1}} f(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} F(\omega) e^{i\omega t} d\omega$$

$$f(t) * g(t) \xrightarrow{\mathcal{F}} F(\omega) G(\omega)$$

$$f(t)g(t) \xrightarrow{\mathcal{F}} F(\omega) * G(\omega)$$

$$\mathcal{F}[f(t) * g(t)] = \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} f(\tau) g(t - \tau) d\tau e^{-i\omega t} dt = \int_{-\infty}^{\infty} f(\tau) \int_{-\infty}^{\infty} g(t - \tau) e^{-i\omega(t - \tau)} dt e^{-i\omega\tau} d\tau$$

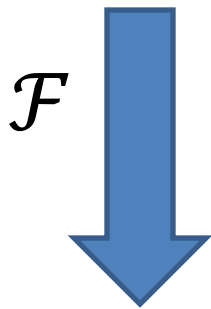
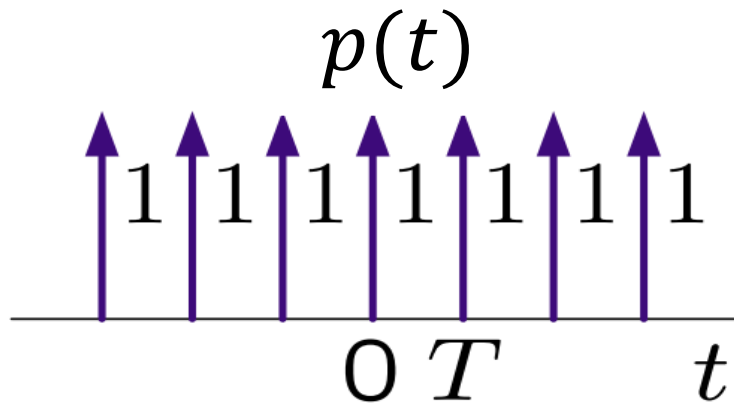
$$= \int_{-\infty}^{\infty} f(\tau) G(\omega) e^{-i\omega\tau} d\tau = G(\omega) \int_{-\infty}^{\infty} f(\tau) e^{-i\omega\tau} d\tau = F(\omega) G(\omega)$$

$$f(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} F(\omega) e^{i\omega t} d\omega \xrightarrow{\substack{t \leftarrow -\omega' \\ \omega \leftarrow t'}} f(-\omega') = \frac{1}{2\pi} \int_{-\infty}^{\infty} F(t') e^{-i\omega' t'} dt' = \frac{1}{2\pi} \mathcal{F}[F(t')] (\omega')$$

$$f(t) \xrightarrow{\mathcal{F}} F(\omega) \Leftrightarrow F(t) \xrightarrow{\mathcal{F}} 2\pi f(-\omega)$$



# Pulse train

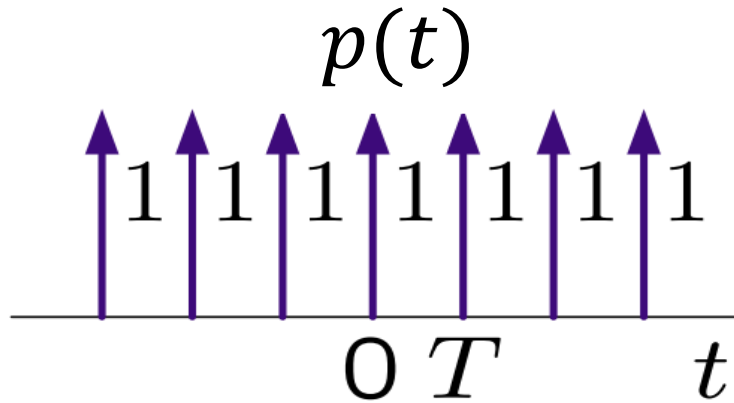


?

- A function  $f(t)$  sampled at  $t = nT$  is simply  $f(t)p(t)$
- $\mathcal{F}[f(t)p(t)] = F(\omega) * P(\omega)$
- What is  $P(\omega)$ ?

$$f(t)g(t) \xrightarrow{\mathcal{F}} F(\omega) * G(\omega)$$

# Fourier Transform of Pulse train



$$p(t) = \sum_{n=-\infty}^{\infty} \delta(t - nT)$$

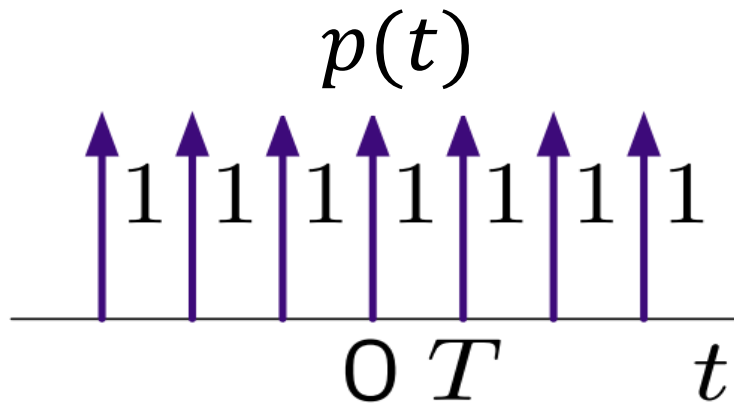
$$\begin{aligned} P(\omega) &= \int_{-\infty}^{\infty} e^{-i\omega t} \sum_{n=-\infty}^{\infty} \delta(t - nT) dt \\ &= \sum_{n=-\infty}^{\infty} \int_{-\infty}^{\infty} e^{-i\omega t} \delta(t - nT) dt = \sum_{n=-\infty}^{\infty} e^{-in\omega T} \end{aligned}$$

$$\begin{aligned} \omega &\leftarrow t \\ T &\leftarrow -\Omega \end{aligned}$$

$$\mathcal{F} \left[ \sum_{n=-\infty}^{\infty} e^{int\Omega} \right] = 2\pi \left[ \sum_{n=-\infty}^{\infty} \delta(\omega + n\Omega) \right] = 2\pi \left[ \sum_{n=-\infty}^{\infty} \delta(\omega - n\Omega) \right]$$

$$f(t) \xrightarrow{\mathcal{F}} F(\omega) \Leftrightarrow F(t) \xrightarrow{\mathcal{F}} 2\pi f(-\omega)$$

# Fourier Series of Pulse train



$$p(t) = \frac{1}{T} \sum_{n=-\infty}^{\infty} e^{i2\pi n \frac{t}{T}}$$

$$\left\langle \sum_{n=-\infty}^{\infty} \delta(t - nT), e^{i2\pi m \frac{t}{T}} \right\rangle = \left\langle \sum_{n=-\infty}^{\infty} c[n] e^{i2\pi n \frac{t}{T}}, e^{i2\pi m \frac{t}{T}} \right\rangle$$

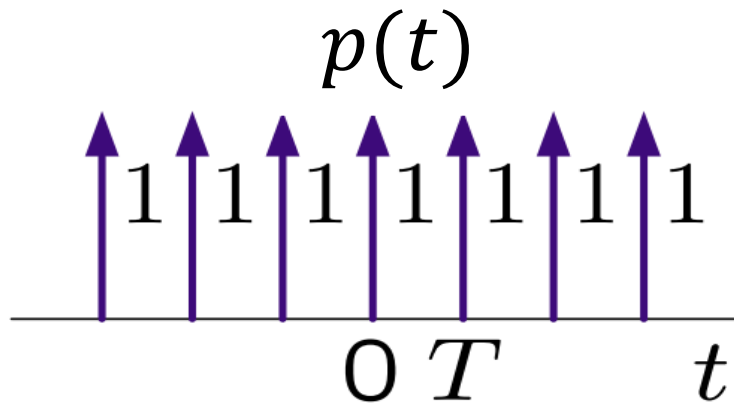
$$p(t) = \sum_{n=-\infty}^{\infty} \delta(t - nT)$$

$$\frac{1}{T} = \frac{1}{T} \int_{-\frac{T}{2}}^{\frac{T}{2}} \sum_{n=-\infty}^{\infty} \delta(t - nT) e^{-i2\pi m \frac{t}{T}} dt = c[m]$$

$$\langle a(t), b(t) \rangle \triangleq \frac{1}{T} \int_{-\frac{T}{2}}^{\frac{T}{2}} a(t) \overline{b(t)} dt$$

$$\left\langle e^{i2\pi n \frac{t}{T}}, e^{i2\pi m \frac{t}{T}} \right\rangle = \frac{1}{T} \int_{-\frac{T}{2}}^{\frac{T}{2}} e^{i2\pi(n-m)\frac{t}{T}} dt = \begin{cases} 1 & \text{if } n = m \\ 0 & \text{otherwise} \end{cases}$$

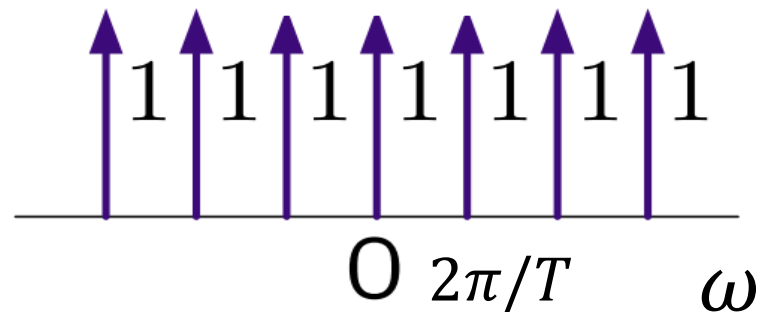
# Fourier Transform of Pulse train



$$p(t) = \frac{1}{T} \sum_{n=-\infty}^{\infty} e^{i2\pi n \frac{t}{T}}$$

$\mathcal{F}$

$P(\omega)$

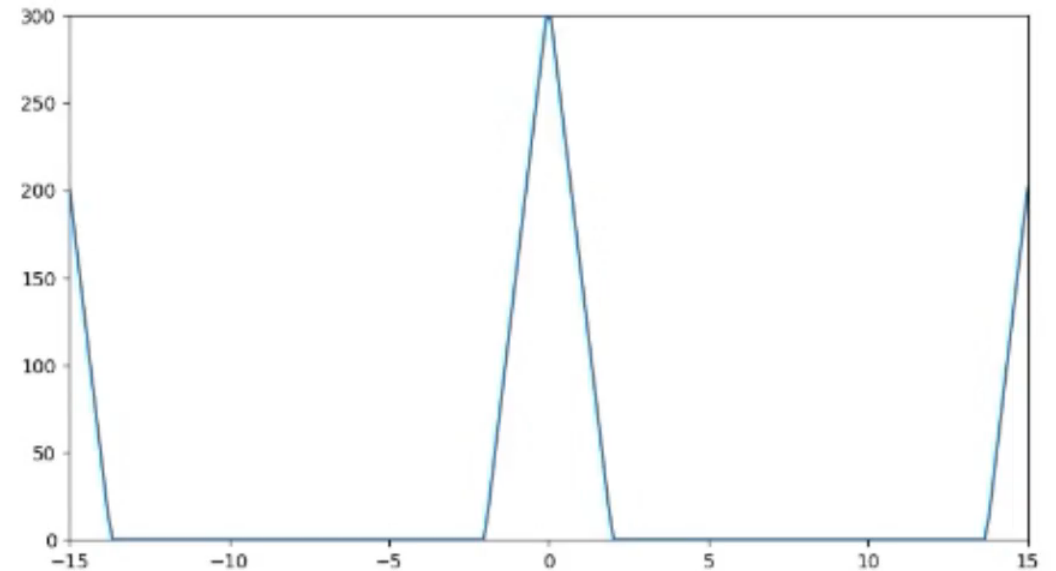
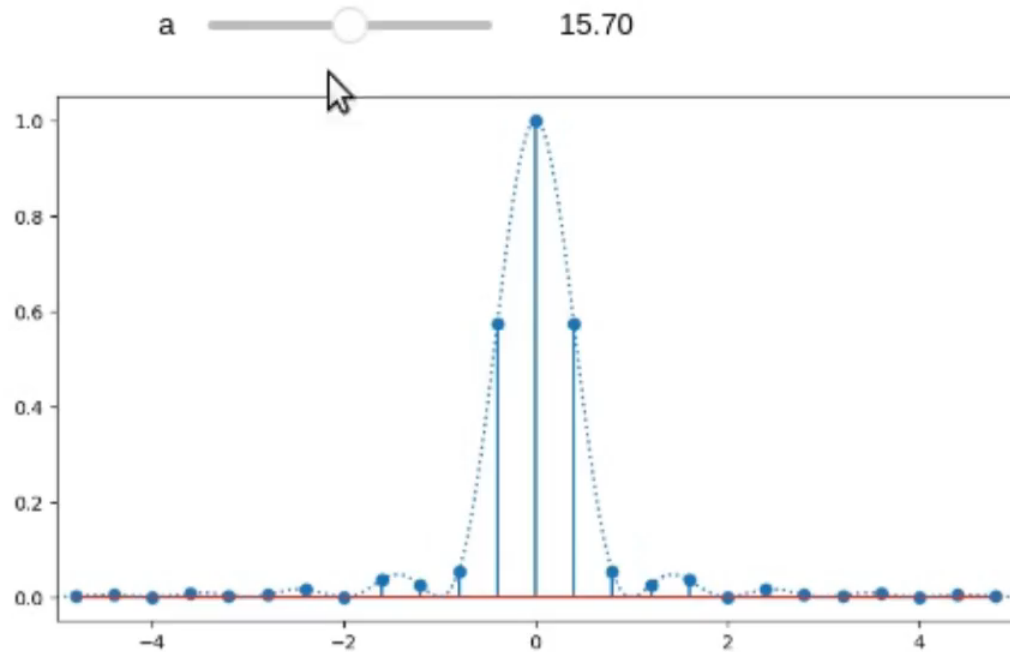


$$\mathcal{F} \left[ \sum_{n=-\infty}^{\infty} \delta(t - nT) \right] = \frac{2\pi}{T} \left[ \sum_{n=-\infty}^{\infty} \delta \left( f - n \left( \frac{2\pi}{T} \right) \right) \right]$$

$$\mathcal{F} \left[ \sum_{n=-\infty}^{\infty} e^{in\Omega t} \right] = 2\pi \left[ \sum_{n=-\infty}^{\infty} \delta(\omega - n\Omega) \right]$$

# Revisit Nyquist-Shannon Theorem

```
interact(plot_sinc,a=widgets.FloatSlider(min=1, max=30, step=0.05, value=15))
```

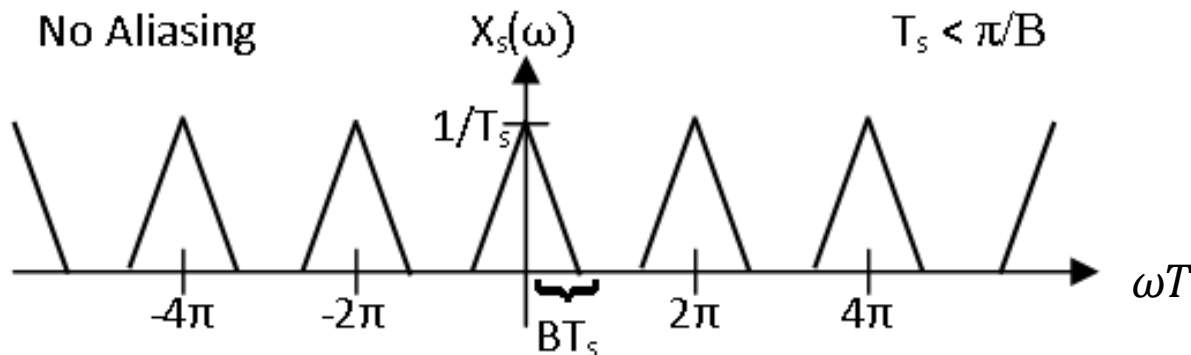
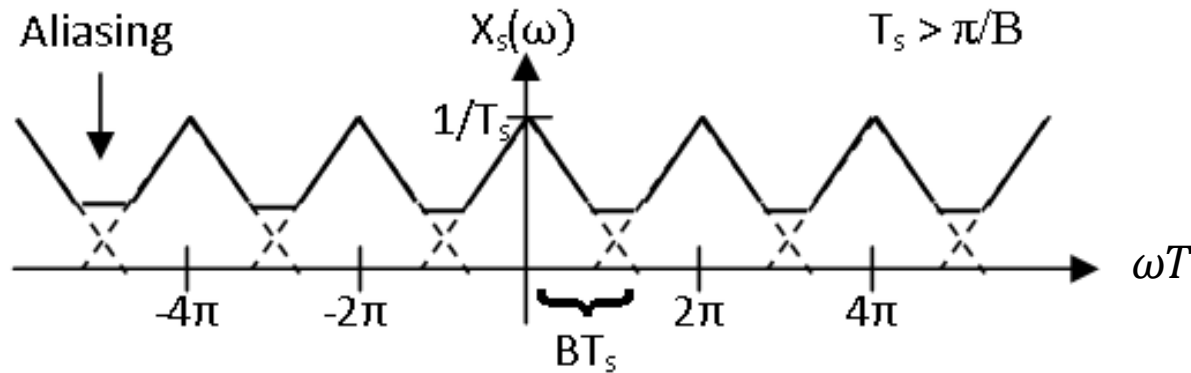
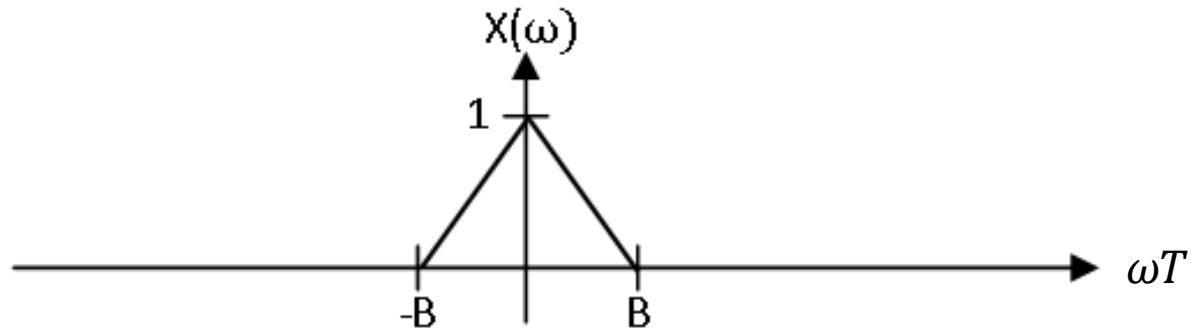


No aliasing as long as bandwidth  $< \frac{2\pi}{T}$

$$\mathcal{F} \left[ \sum_{n=-\infty}^{\infty} \delta(t - nT) \right] = \frac{2\pi}{T} \left[ \sum_{n=-\infty}^{\infty} \delta \left( \omega - n \left( \frac{2\pi}{T} \right) \right) \right]$$

$$f(t)g(t) \xrightarrow{\mathcal{F}} F(\omega) * G(\omega)$$

# Aliasing in downsampling



- Downsampling is just resampling at lower rate
- Aliasing if baseband overlaps

$$\mathcal{F} \left[ \sum_{n=-\infty}^{\infty} \delta(t - nT) \right] = \frac{2\pi}{T} \left[ \sum_{n=-\infty}^{\infty} \delta \left( \omega - n \left( \frac{2\pi}{T} \right) \right) \right]$$

$$f(t)g(t) \xrightarrow{\mathcal{F}} F(\omega) * G(\omega)$$

# Anti-aliasing

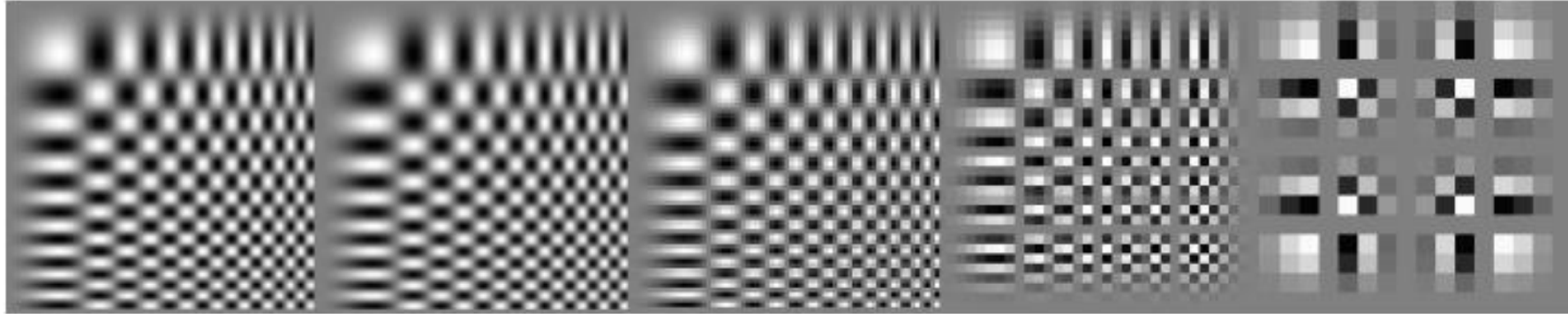
256x256

128x128

64x64

32x32

16x16



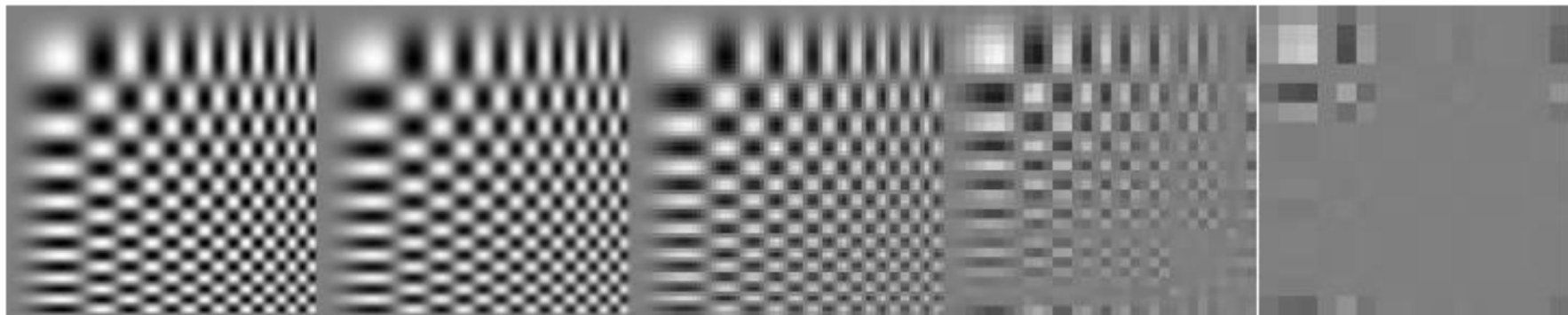
256x256

128x128

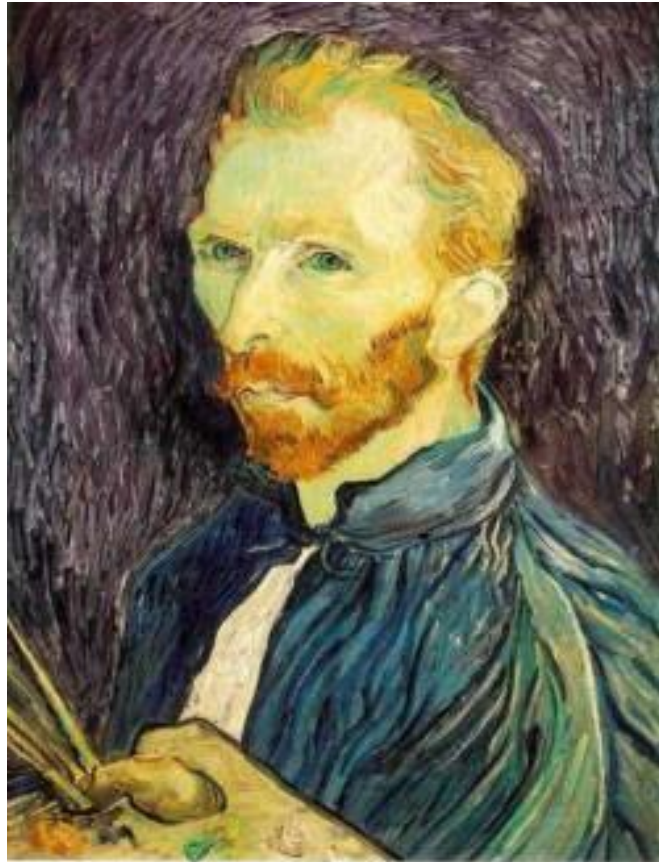
64x64

32x32

16x16



# Gaussian pre-filtering



Gaussian 1/2



G 1/4

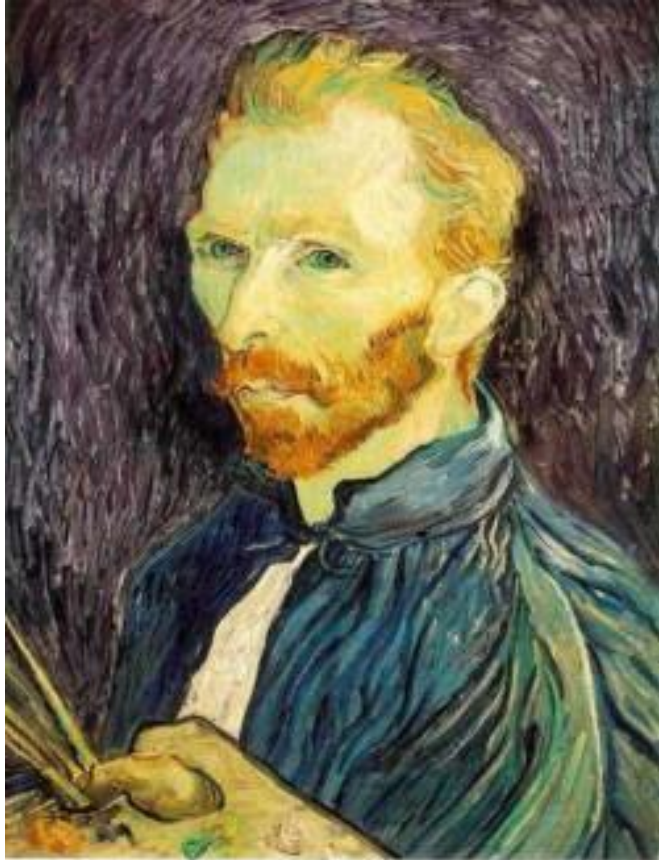


G 1/8

- Solution: filter the image, *then* subsample



# Subsampling with Gaussian pre-filtering



Gaussian 1/2



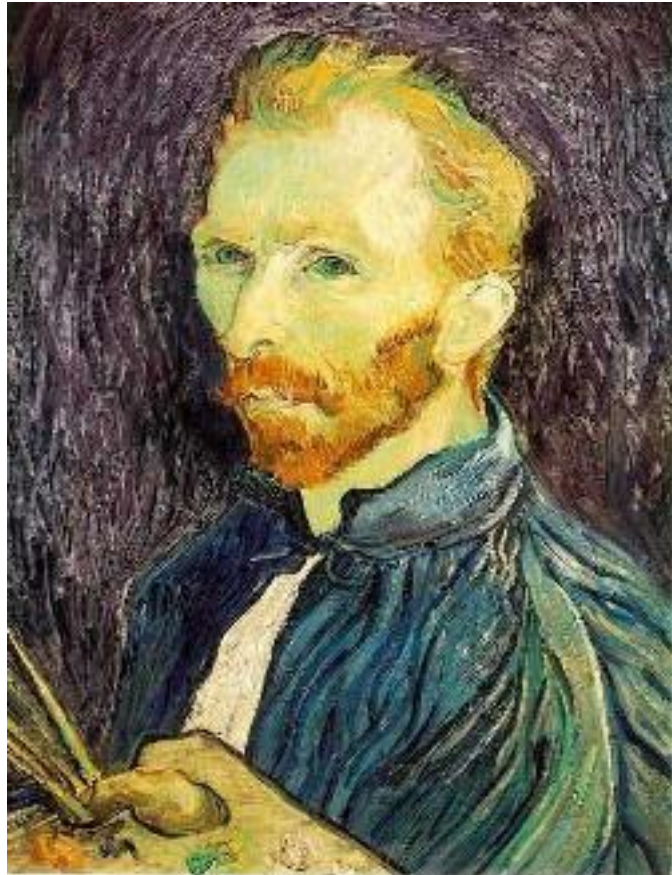
G 1/4



G 1/8

- Solution: filter the image, *then* subsample

# Compare with...



1/2




1/4 (2x zoom)



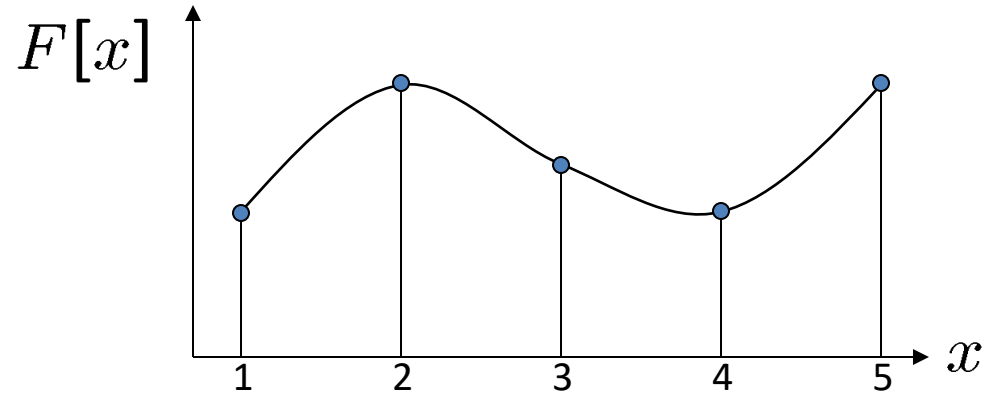
1/8 (4x zoom)

# Upsampling

- This image is too small for this screen: 
- How can we make it 10 times as big?
- Simplest approach:
  - repeat each row
  - and column 10 times
- (“Nearest neighbor interpolation”)



# Image interpolation



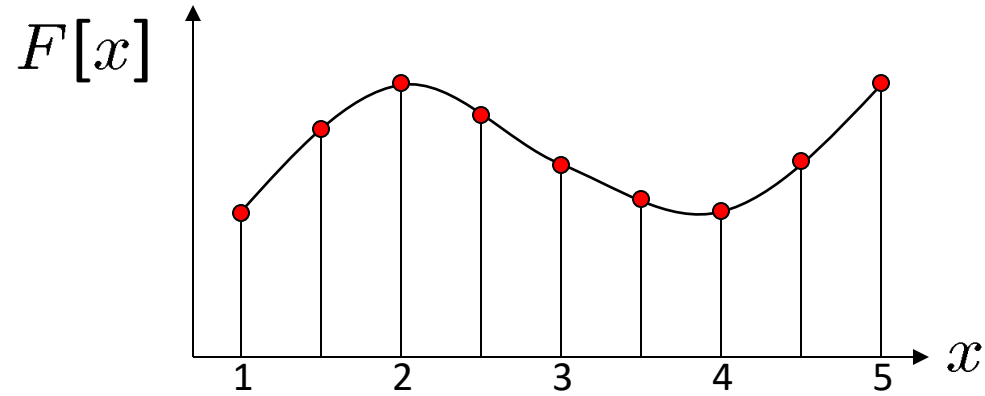
$d = 1$  in this example

Recall how a digital image is formed

$$F[x, y] = \text{quantize}\{f(xd, yd)\}$$

- It is a discrete point-sampling of a continuous function
- If we could somehow reconstruct the original function, any new image could be generated, at any resolution and scale

# Image interpolation



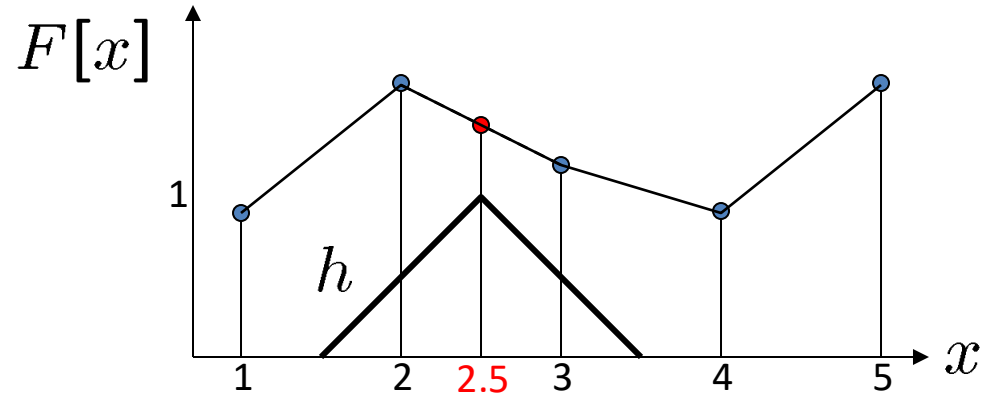
$d = 1$  in this example

Recall how a digital image is formed

$$F[x, y] = \text{quantize}\{f(xd, yd)\}$$

- It is a discrete point-sampling of a continuous function
- If we could somehow reconstruct the original function, any new image could be generated, at any resolution and scale

# Image interpolation



- What if we don't know  $f$  ?

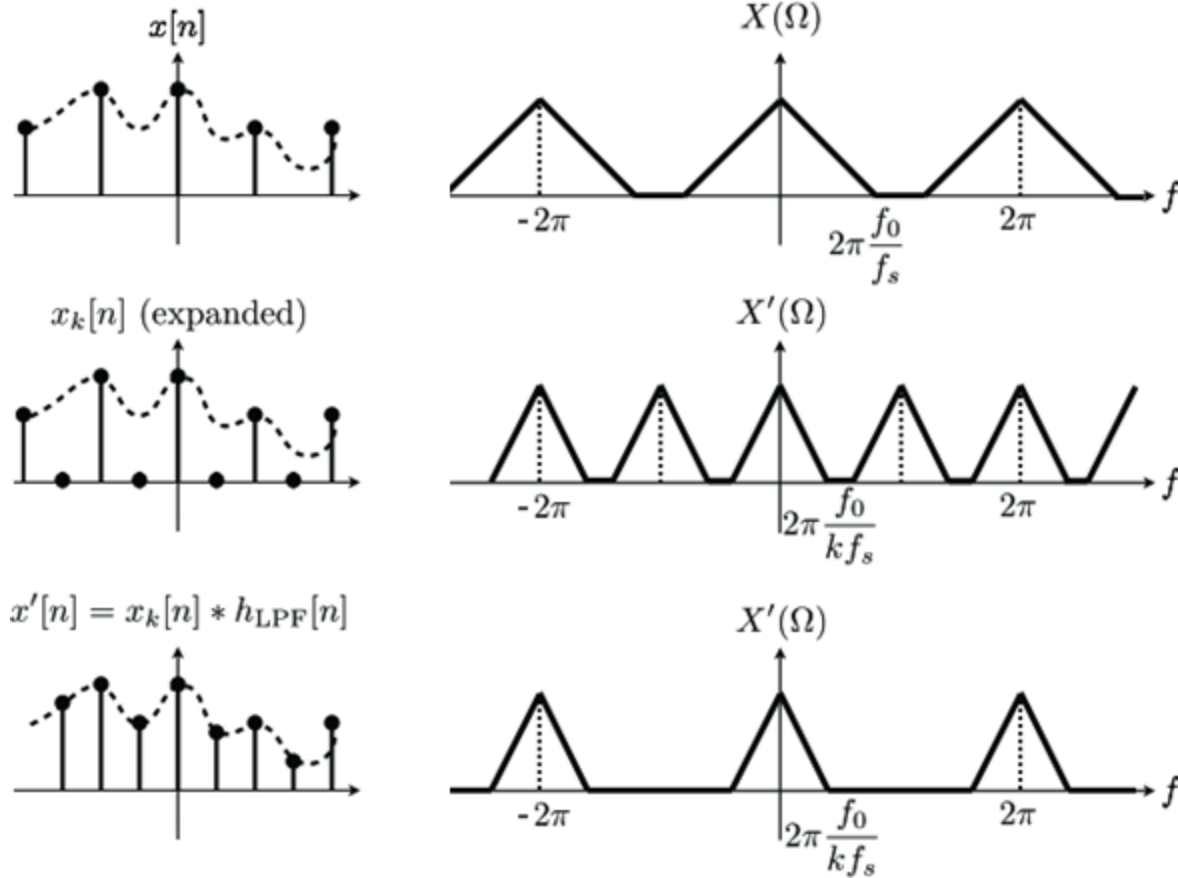
- Guess an approximation:  $\tilde{f}$
- Can be done in a principled way: filtering
- Convert  $F$  to a continuous function:

$$f_F(x) = F\left(\frac{x}{d}\right) \text{ when } \frac{x}{d} \text{ is an integer, } 0 \text{ otherwise}$$

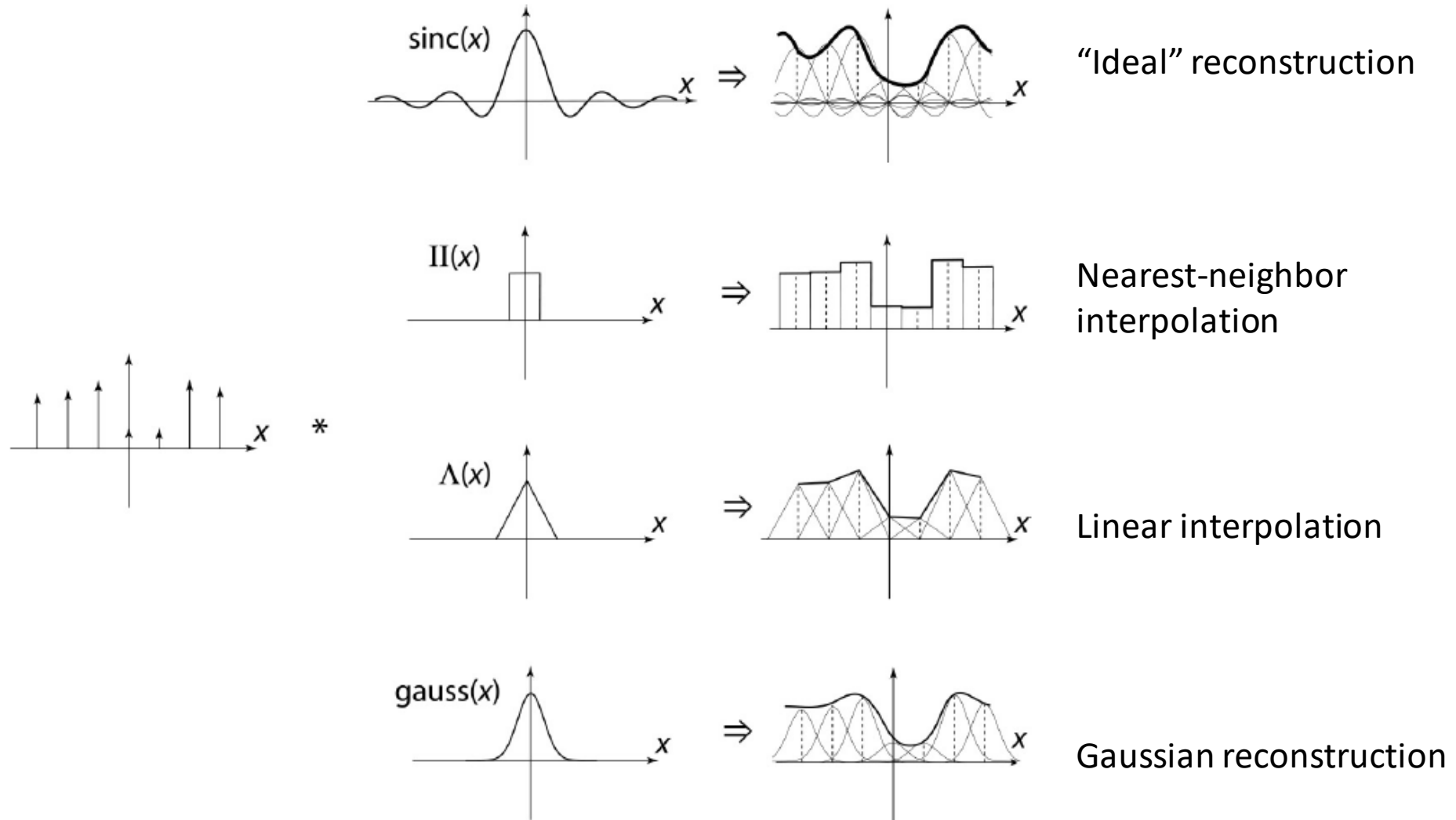
- Reconstruct by convolution with a *reconstruction filter*,  $h$

$$\tilde{f} = h * f_F$$

# Frequency representation



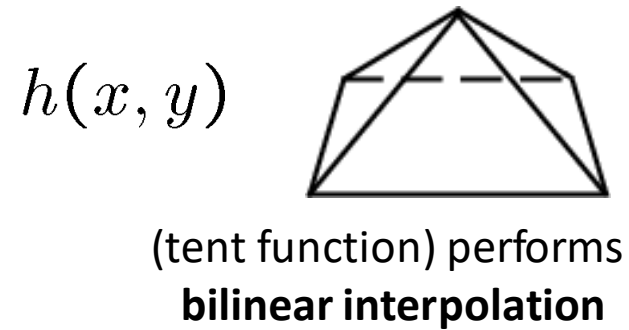
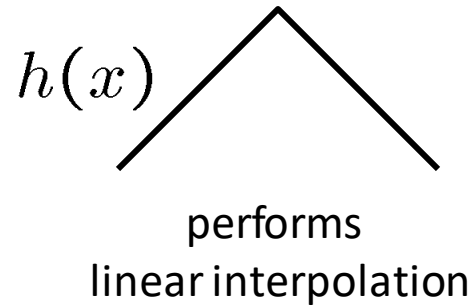
# Image interpolation





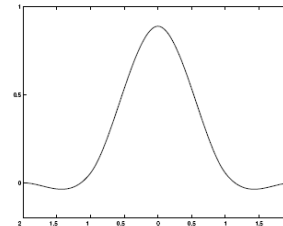
# Reconstruction filters

- What does the 2D version of this hat function look like?



Better filters give better resampled images

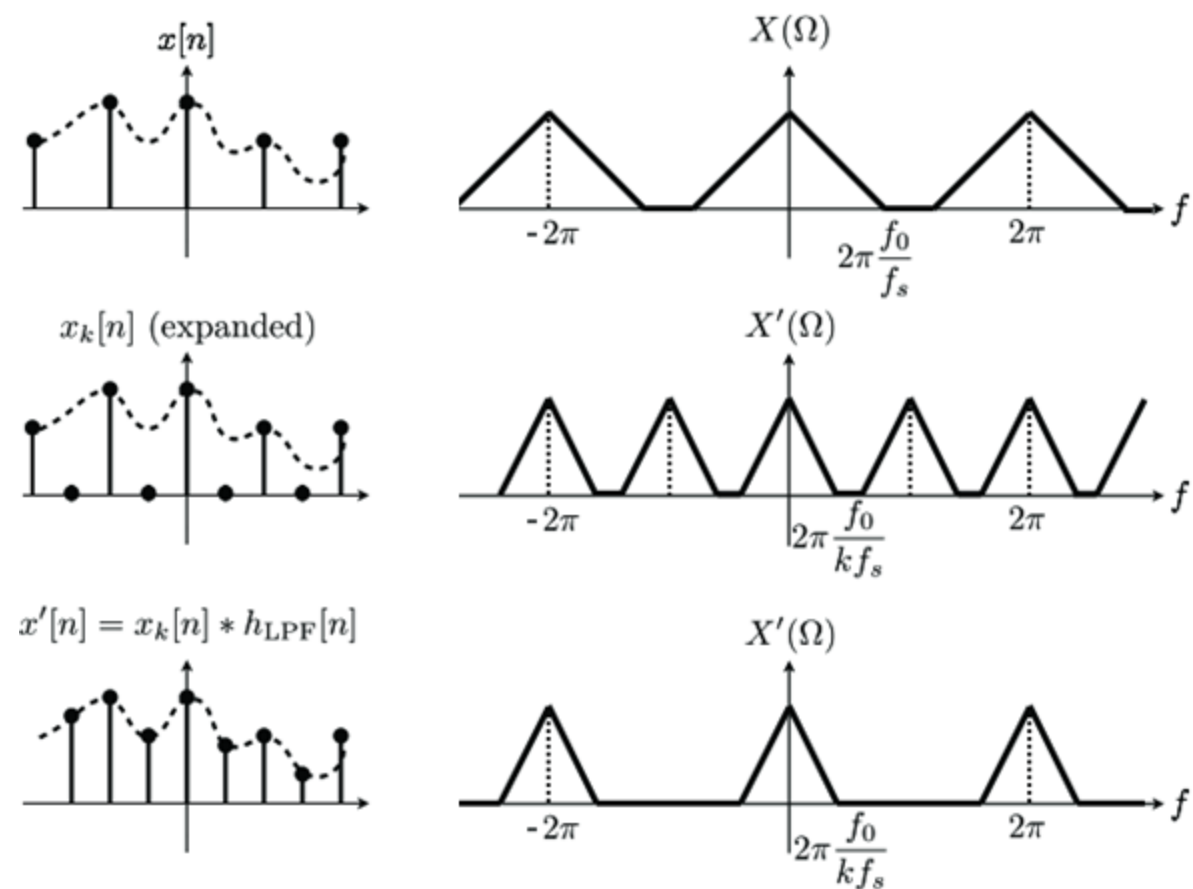
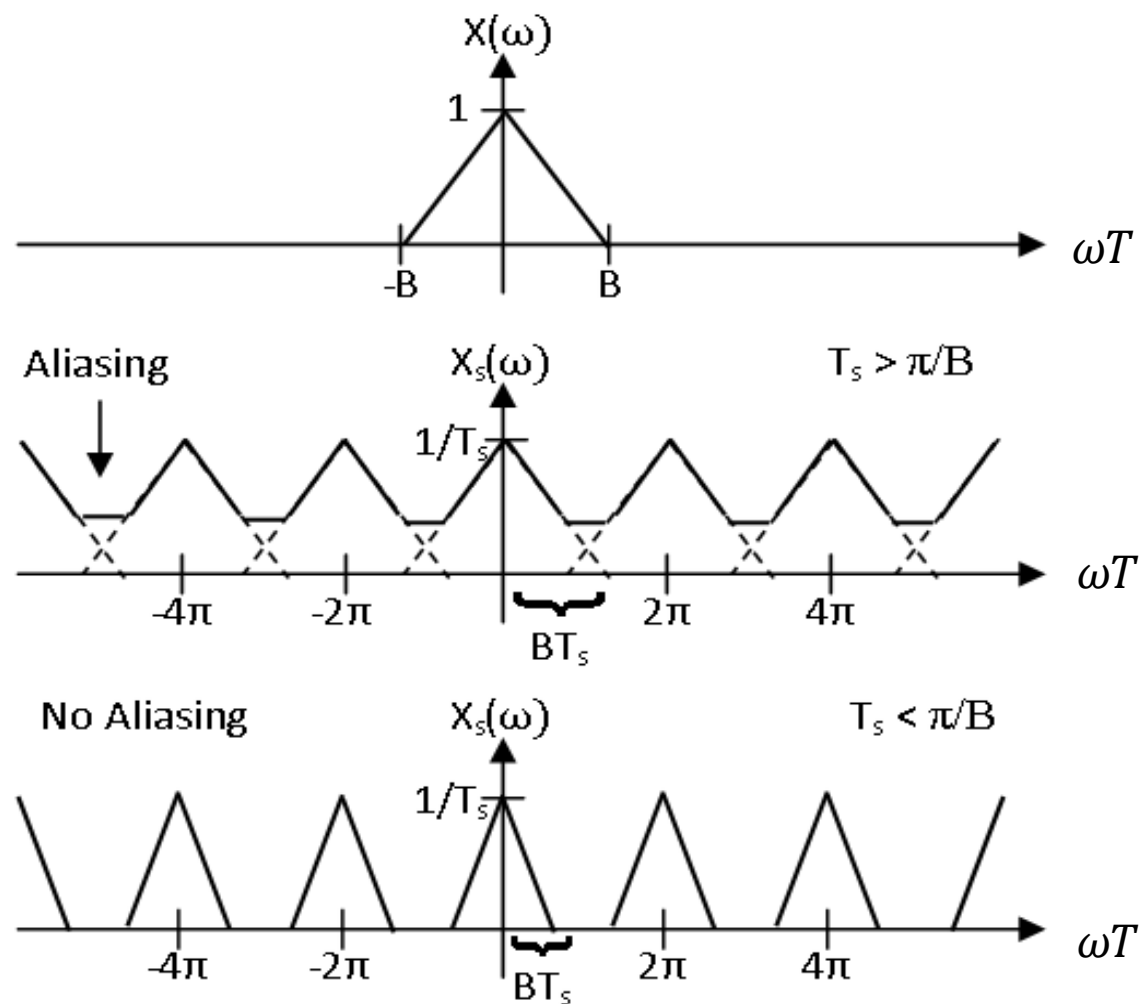
- **Bicubic** is common choice



Cubic reconstruction filter

$$r(x) = \frac{1}{6} \begin{cases} (12 - 9B - 6C)|x|^3 + (-18 + 12B + 6C)|x|^2 + (6 - 2B) & |x| < 1 \\ ((-B - 6C)|x|^3 + (6B + 30C)|x|^2 + (-12B - 48C)|x| + (8B + 24C)) & 1 \leq |x| < 2 \\ 0 & \text{otherwise} \end{cases}$$

# Summary: downsampling and upsampling

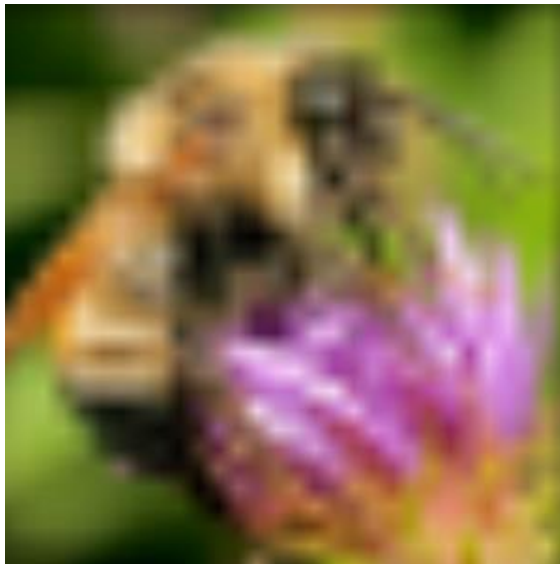


# Image interpolation

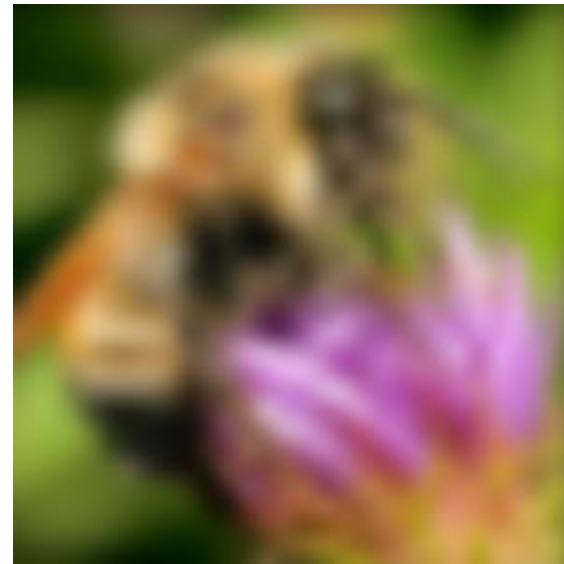
Original image:  x 10



Nearest-neighbor interpolation

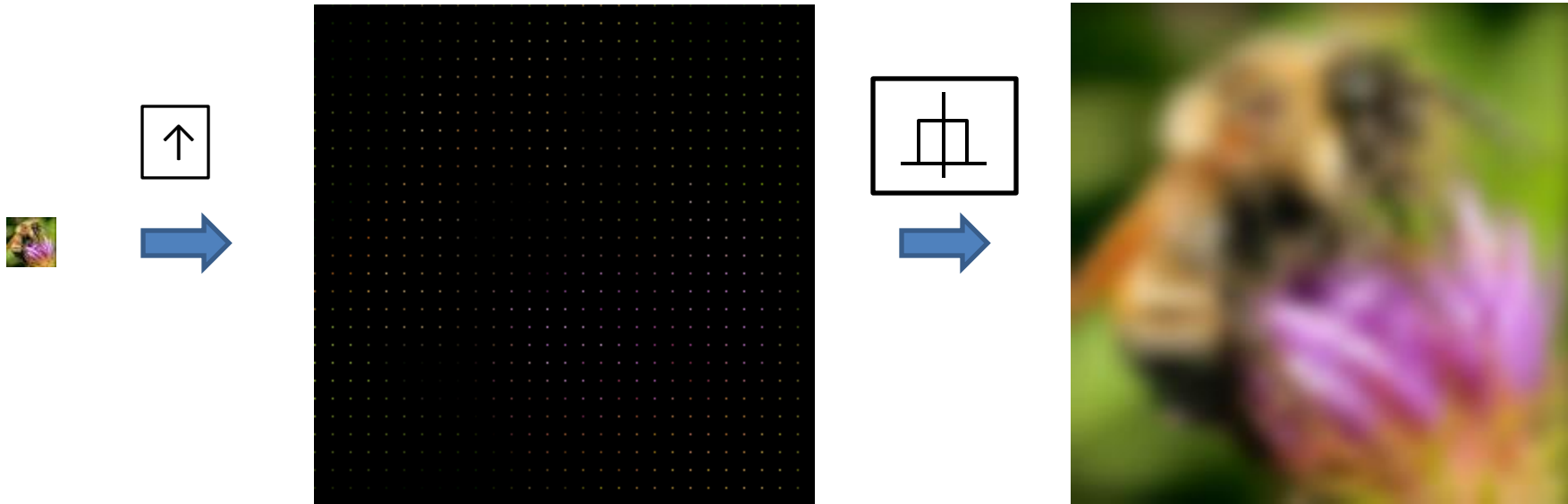


Bilinear interpolation

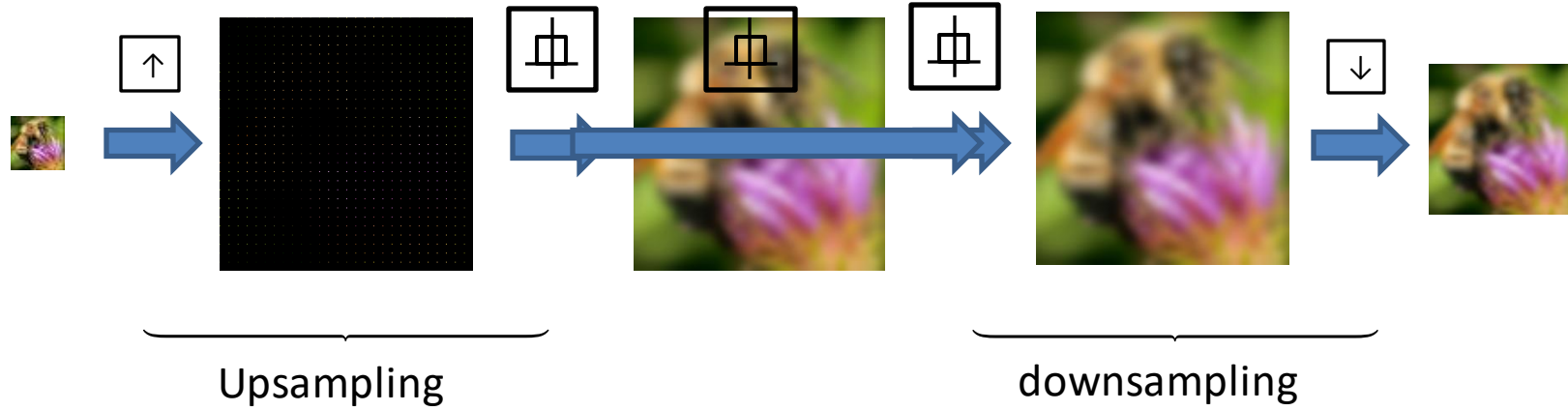


Bicubic interpolation

# DSP Interpretation



# Image resampling



# Hybrid Image

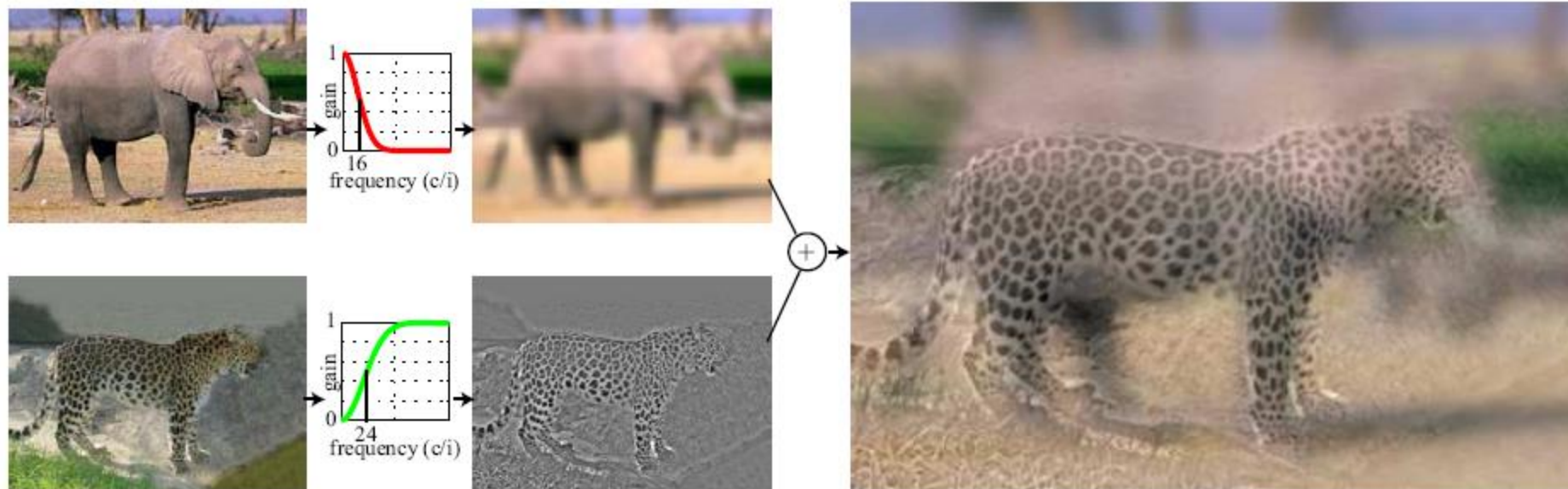


Salvador Dali, 1976

# Another example

- Who is (s)he?

# Hybrid Images

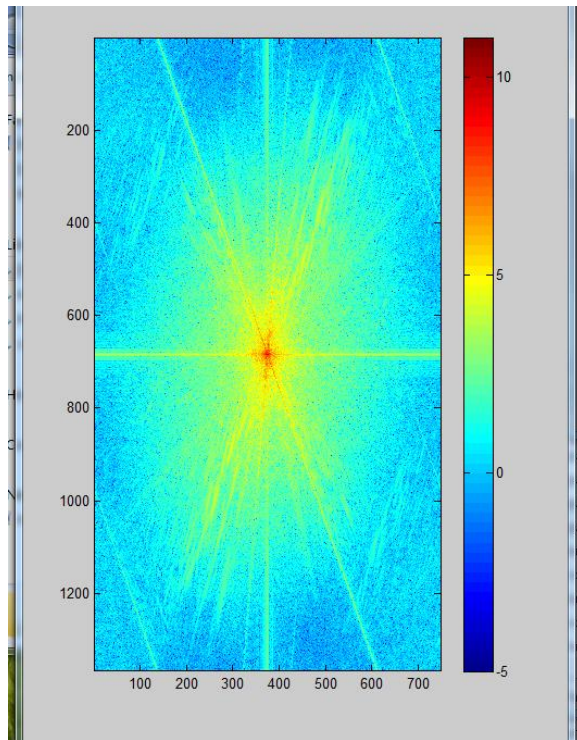


- A. Oliva, A. Torralba, P.G. Schyns, [“Hybrid Images,”](#) SIGGRAPH 2006



# Hybrid Image in FFT

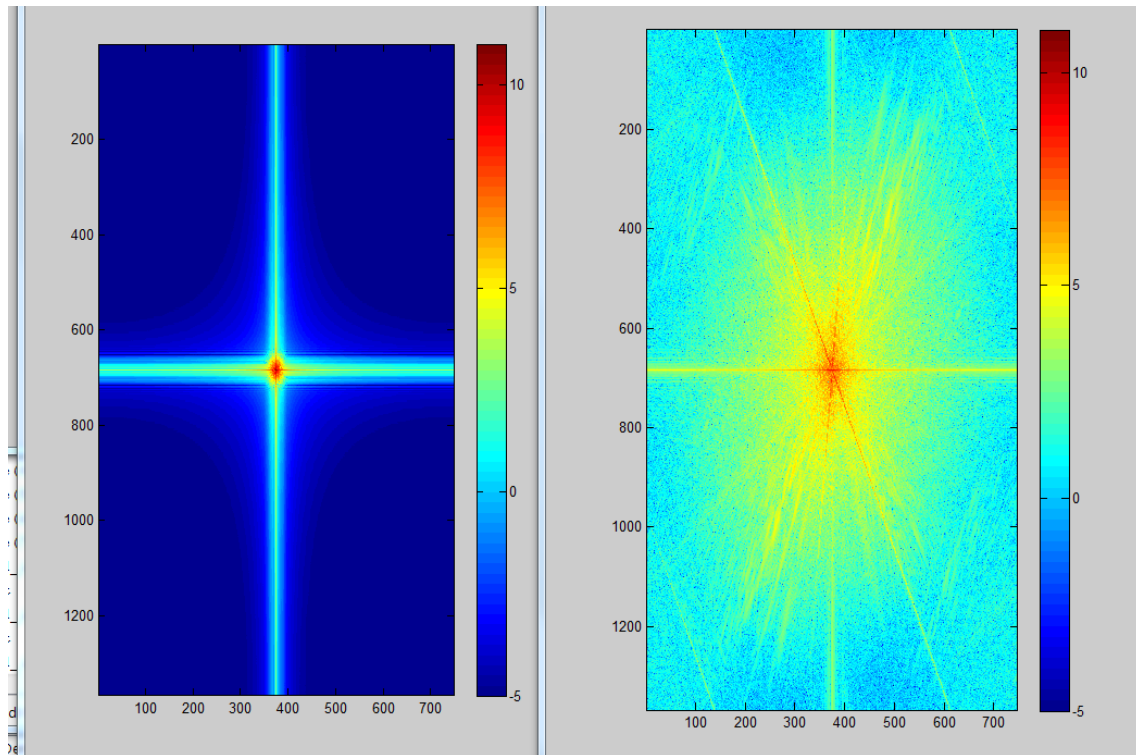
Hybrid Image



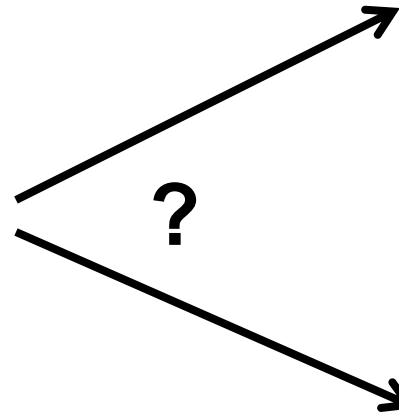
Low-passed Image



High-passed Image

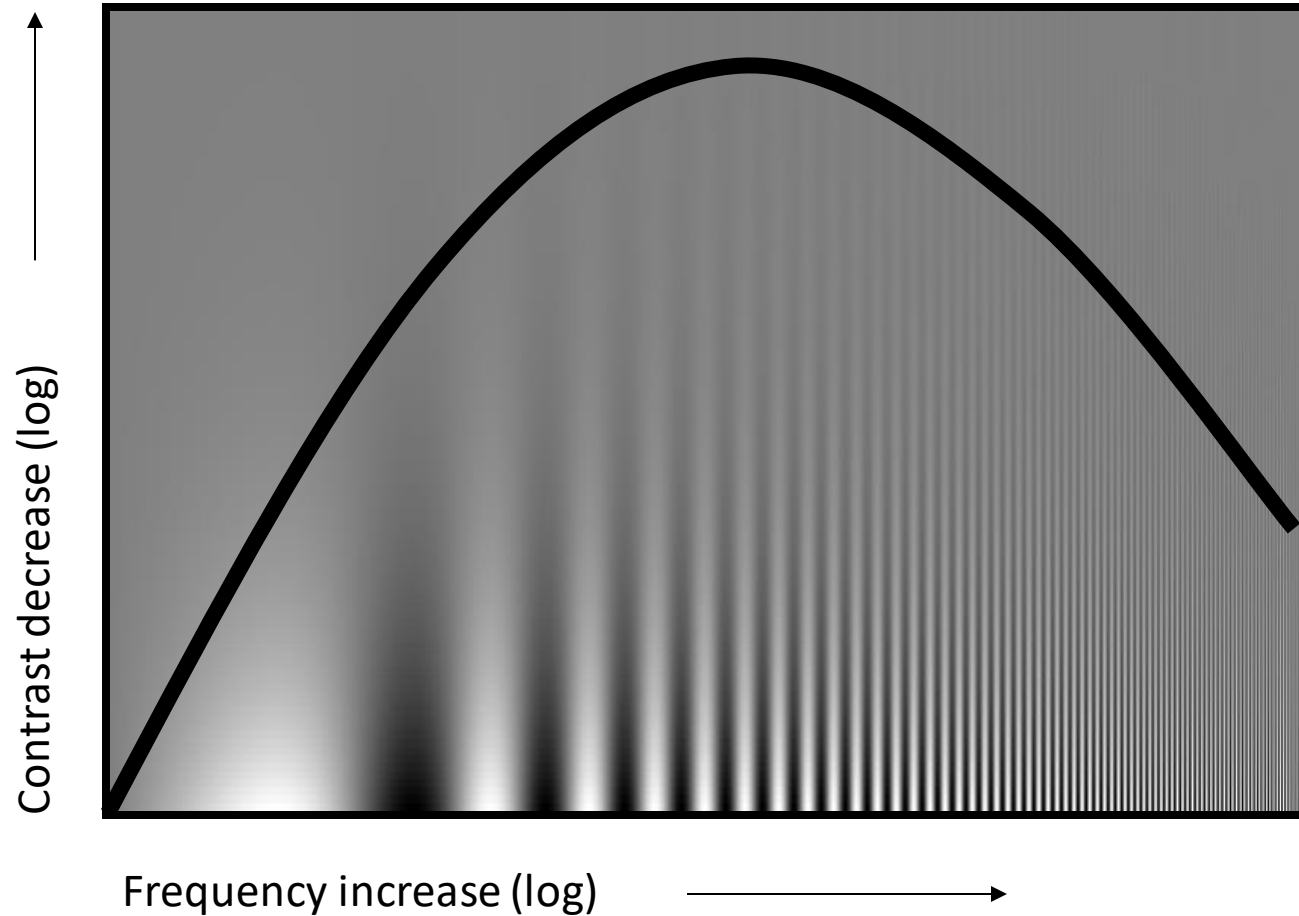


# Why do we get different, distance-dependent interpretations of hybrid images?

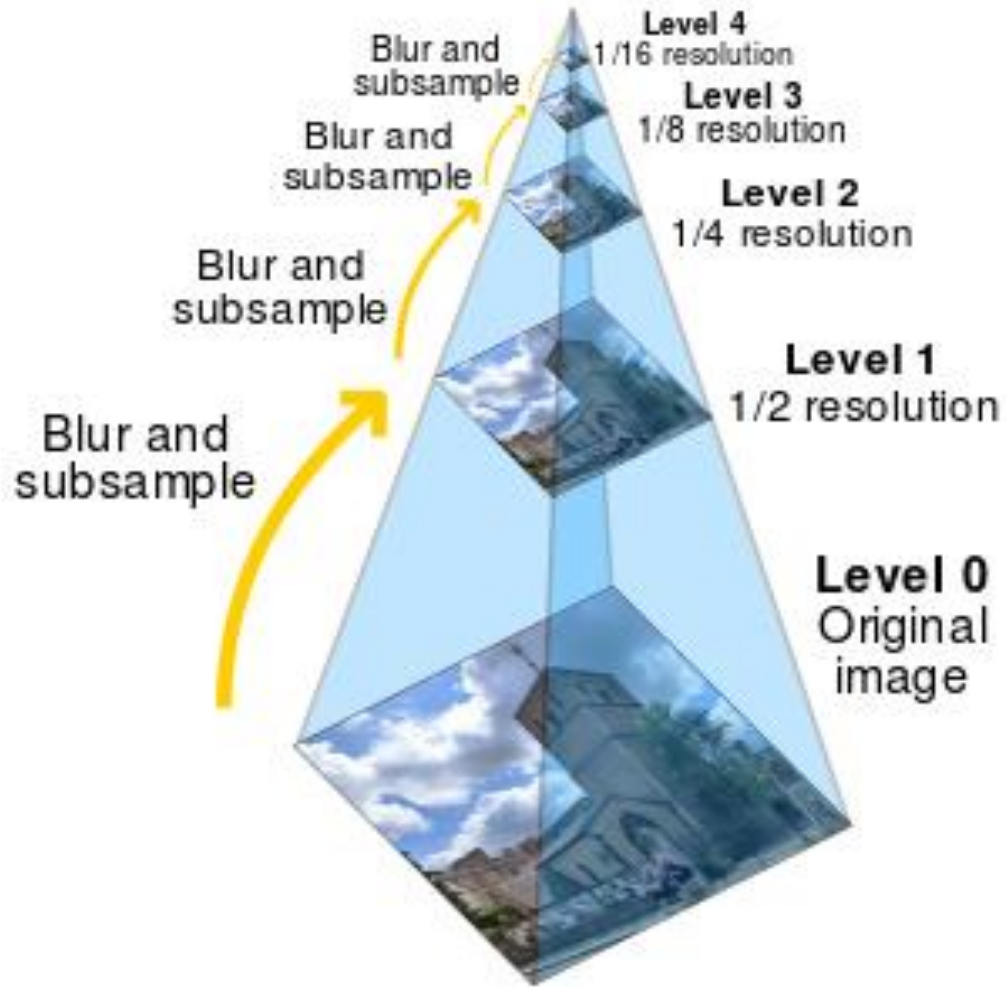


# Campbell-Robson contrast sensitivity curve

Perceptual cues in the mid-high frequencies dominate perception.

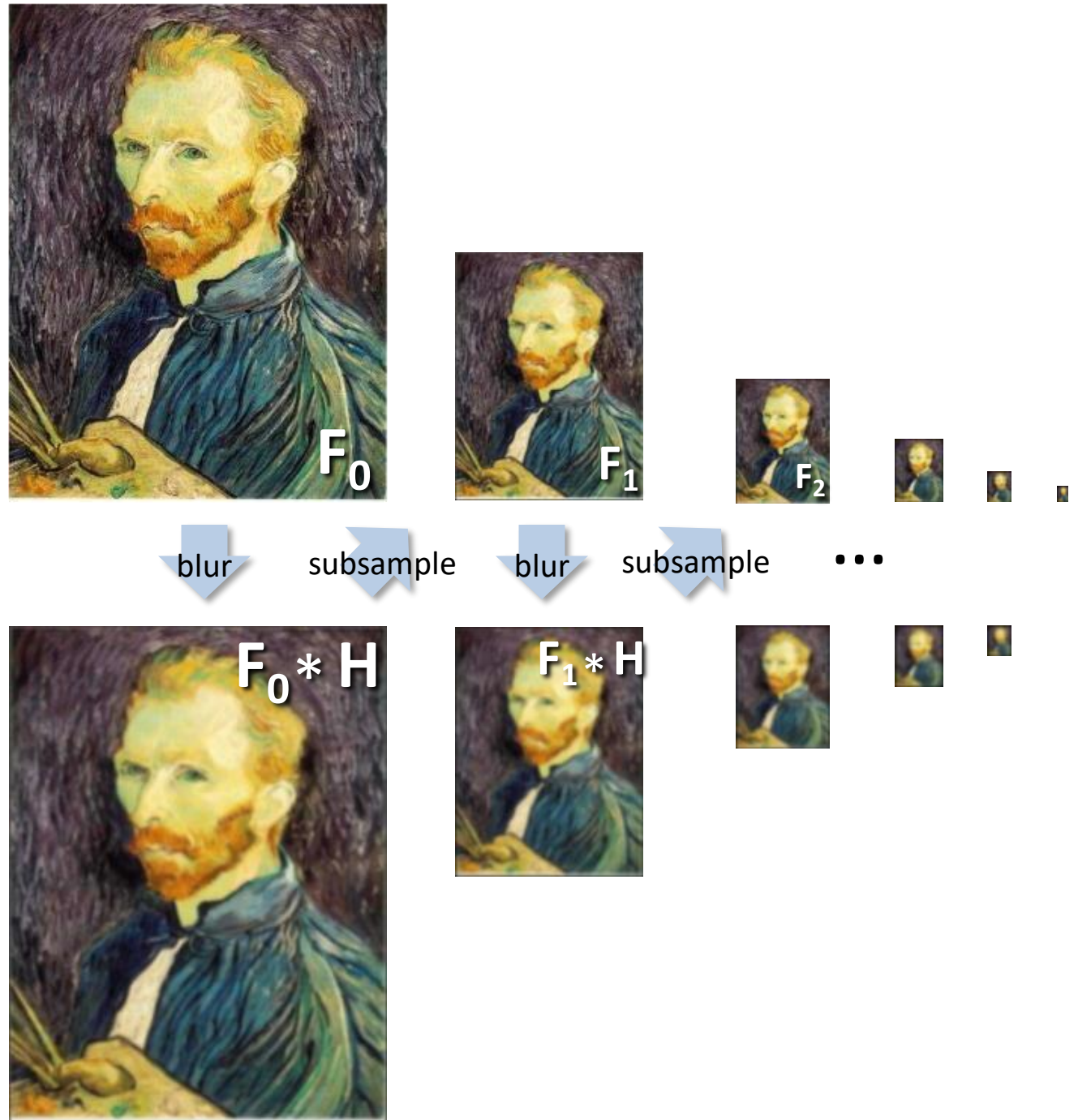


# Image Pyramids

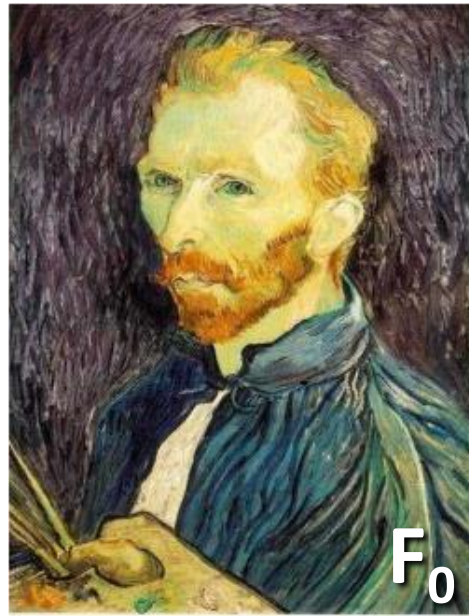
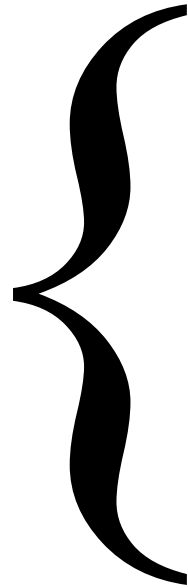


Project 1 function:  
`vis_hybrid_image.m`

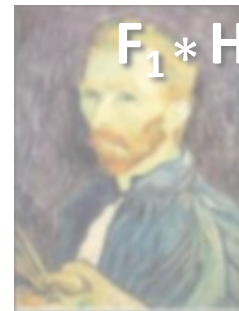
# Gaussian pyramid



*Gaussian pyramid*

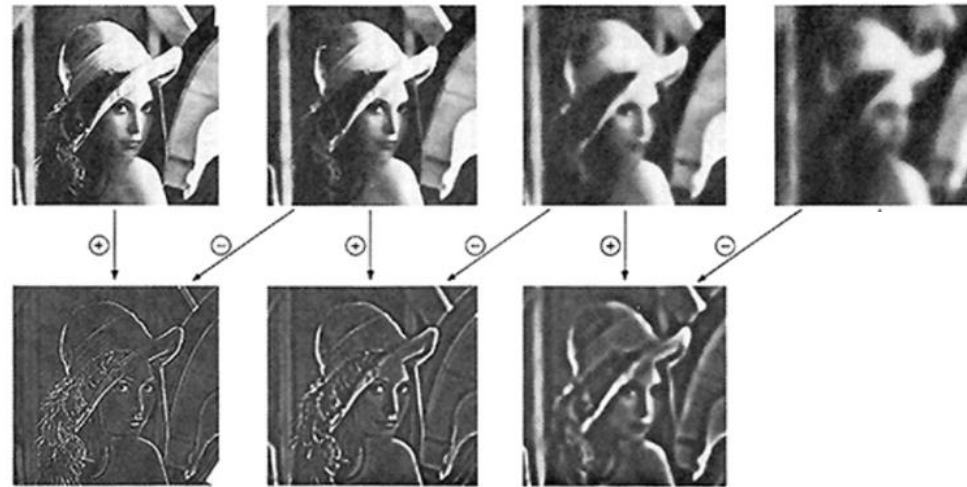


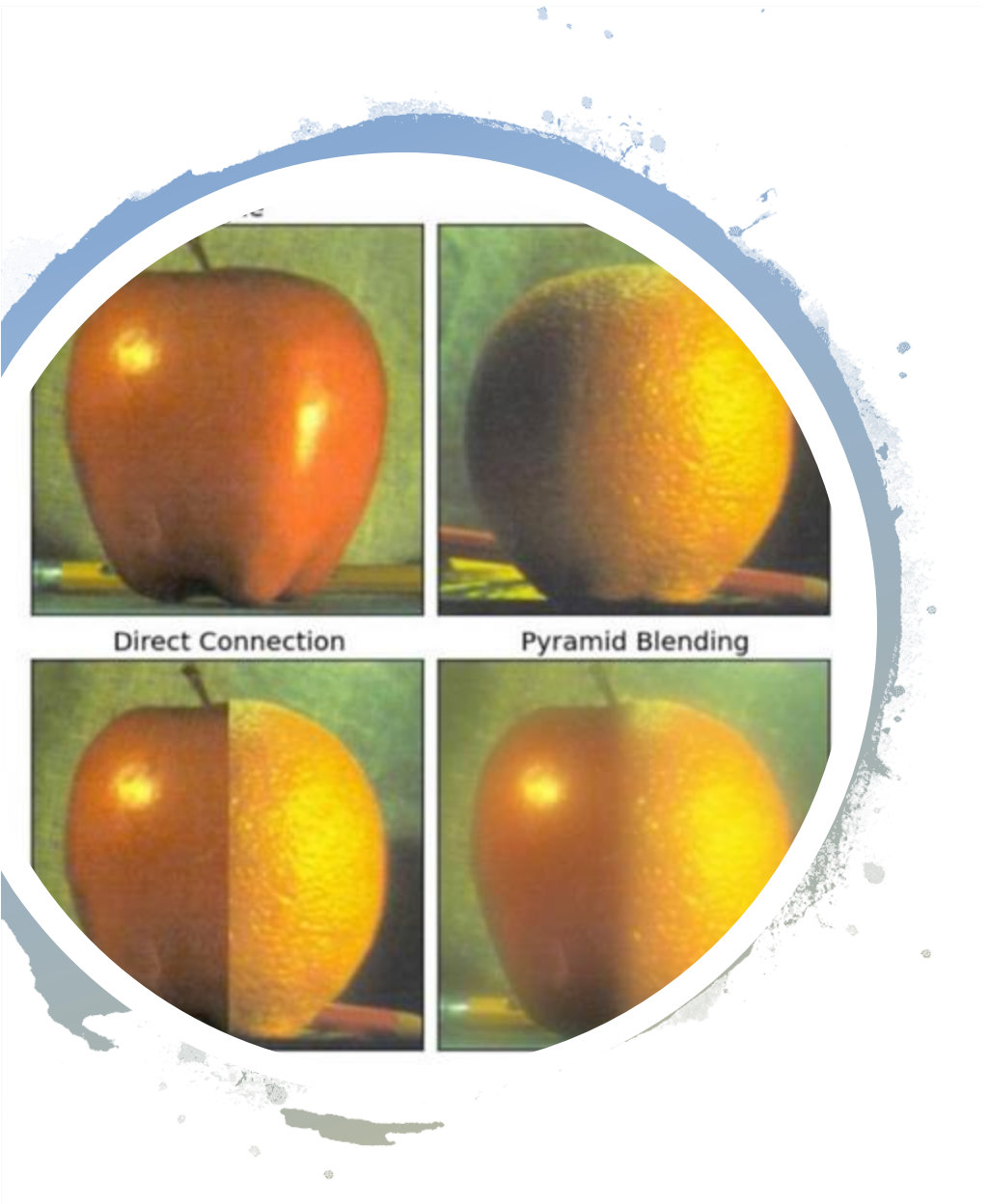
...



# Laplace Pyramid

- Derive from Gaussian pyramid
  - $G1 = \text{pydn}(G0)$ ;  $G2 = \text{pydn}(G1)$ , ...
  - One level of laplace pyramid is difference between approximated and original Gaussian pyramid levels
  - $L0 = G0 - \text{pyup}(G1)$ ;  $L1 = G1 - \text{pyup}(G2)$





# Image compositing

- Generate L-pyramid of orange
- Generate L-pyramid of apple
- Combine two pyramids
  - For all levels, one half from one pyramid, the other half from another
- Reconstruct image from combine pyramid



# Summary

- Product in time domain = convolution in freq domain
  - Sampling can be represented as signal multiplied by pulse train
  - Infinite repeated copy in frequency domain
  - When copies overlaps => aliasing
- Downsampling naively will lead to **aliasing**
  - Solution: apply low pass filter before downsample
- Should apply low pass filter after upsampling
- Laplace pyramid and Gaussian pyramid
- Hybrid image and compositing image