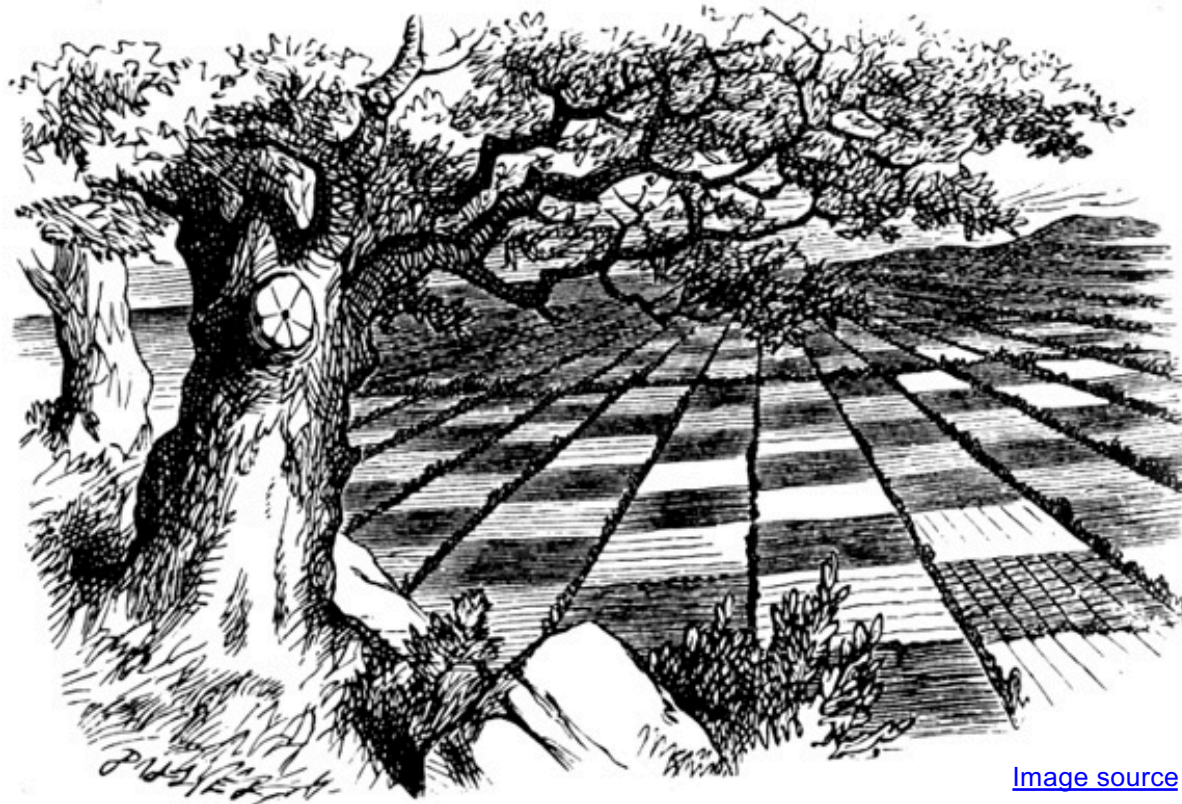


# A gentle introduction to Fourier analysis

---



[Image source](#)

Many slides borrowed from S. Seitz, A. Efros, D. Hoiem, B. Freeman, A. Zisserman

# Mystery 1

---

- Why can downsampling sometimes lead to aliasing?

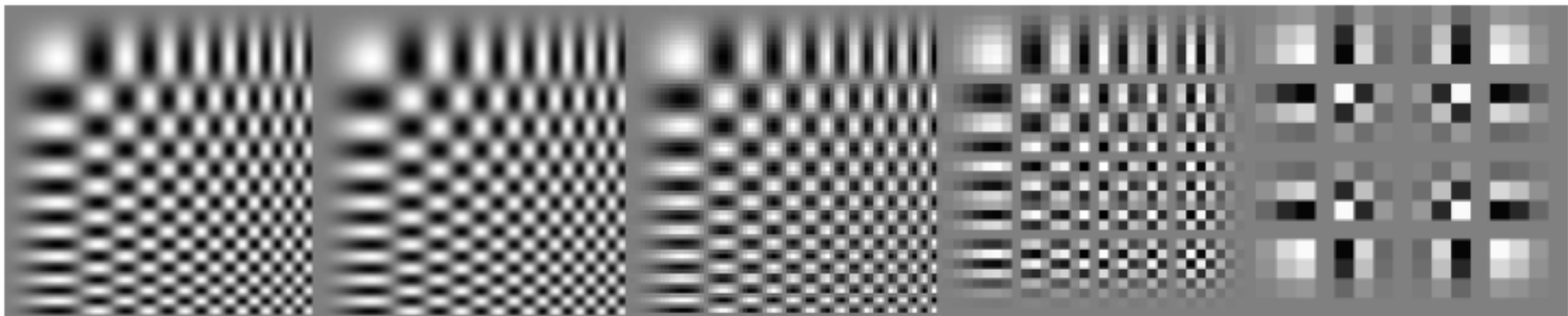
256x256

128x128

64x64

32x32

16x16

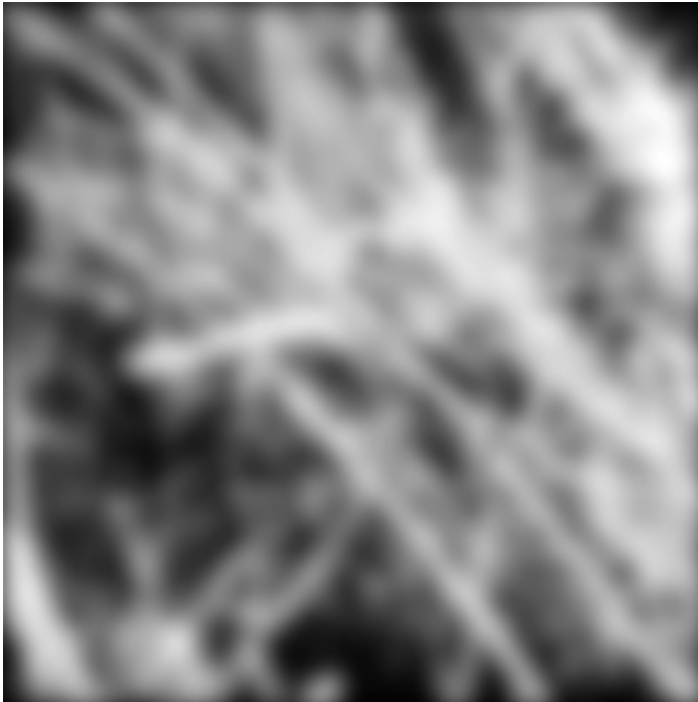


## Mystery 2

---

- Why does filtering with a Gaussian give a nice smooth image, but filtering with a box filter gives artifacts?

Gaussian



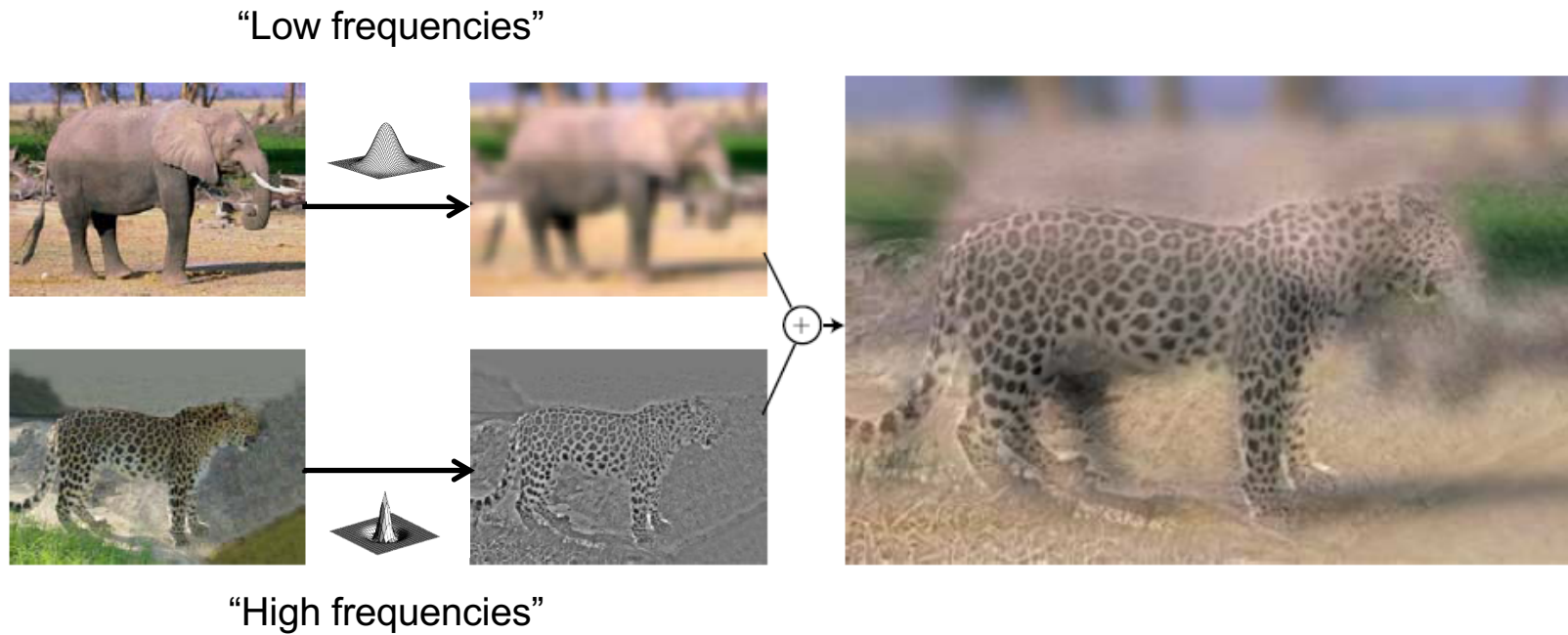
Box filter



# Mystery 3

---

- How do hybrid images work?

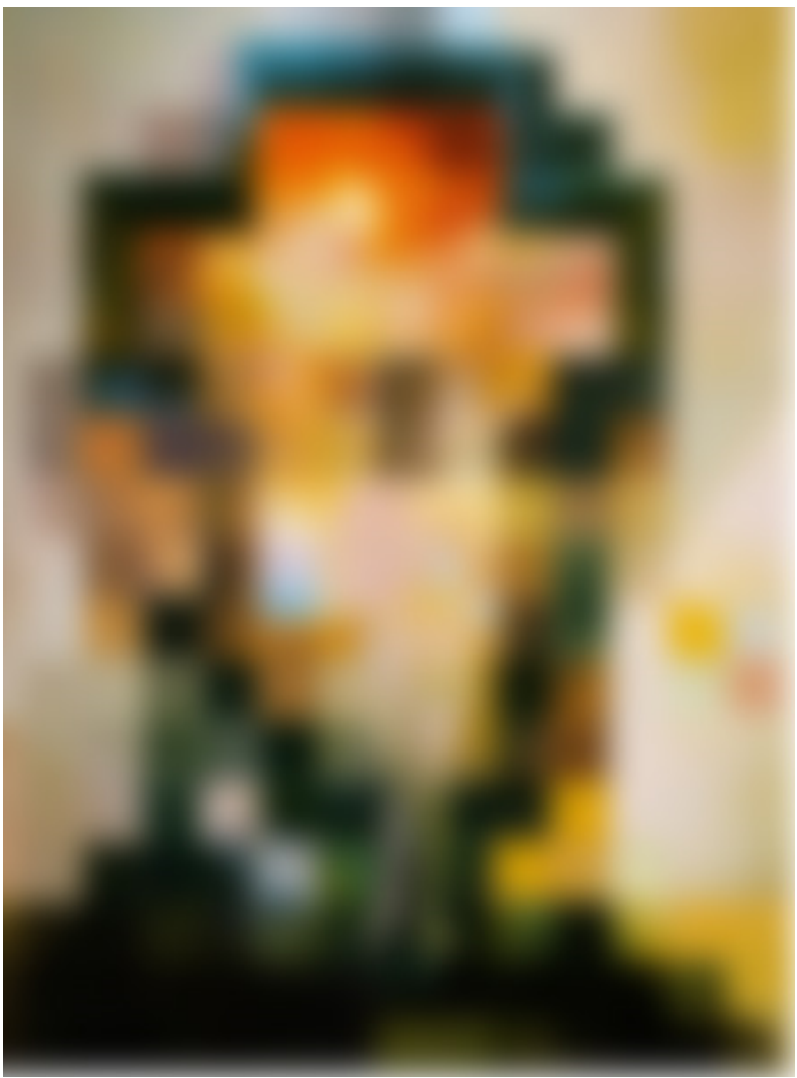


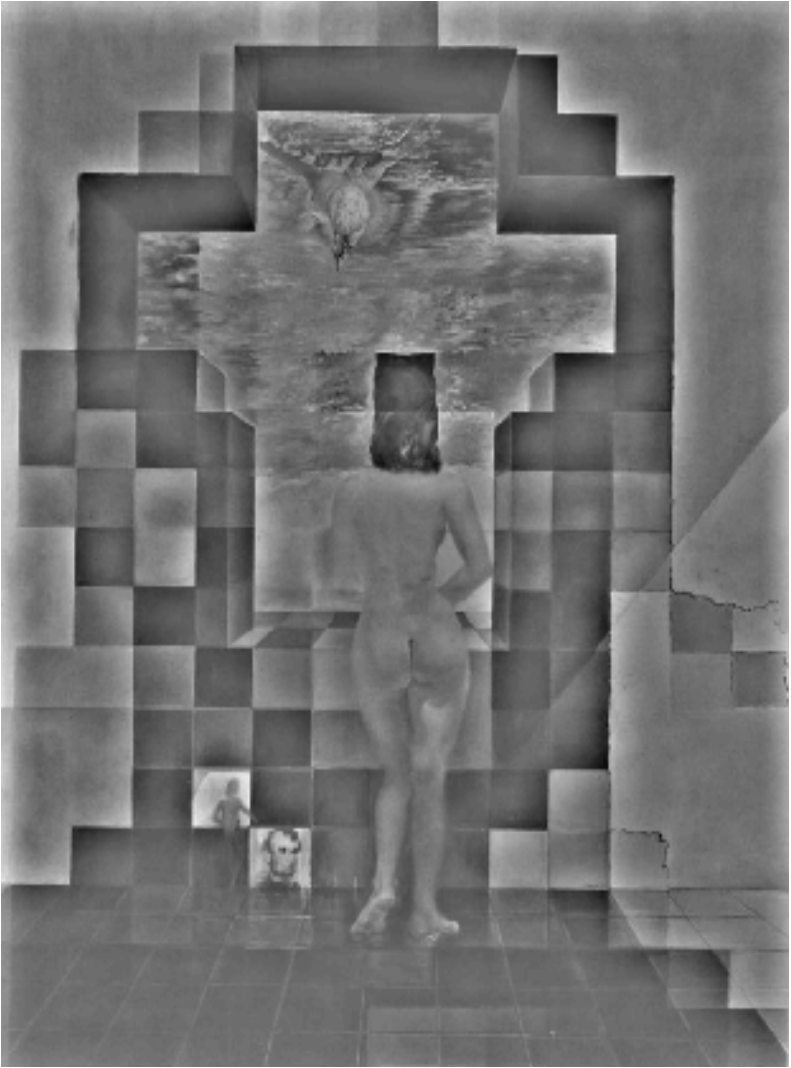
A. Oliva, A. Torralba, P.G. Schyns, [Hybrid Images](#), SIGGRAPH 2006



**Salvador Dali**  
*"Gala Contemplating the Mediterranean Sea,*  
*which at 30 meters becomes the portrait*  
*of Abraham Lincoln", 1976*



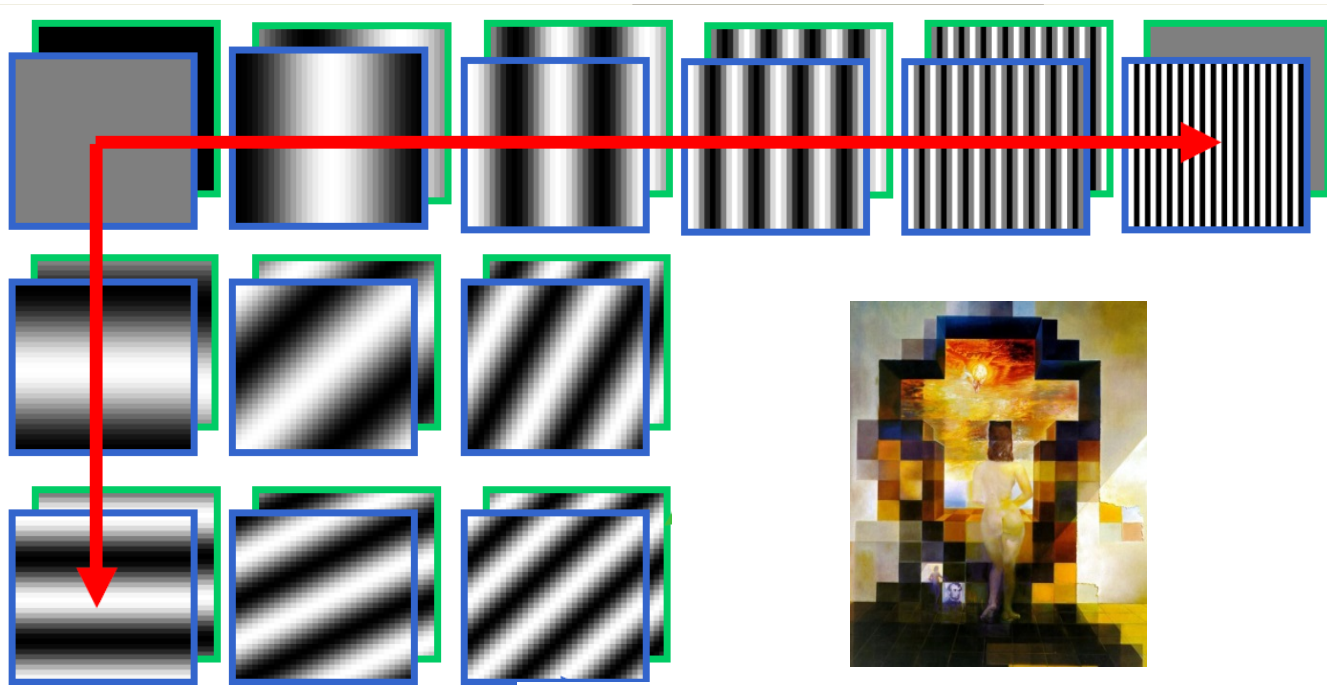




# Fourier analysis

---

- To understand such phenomena, we need a representation of images that allows us to tease apart slow and fast changes



# Outline

---

- 1D Fourier transform
  - Definition and properties
  - Discrete Fourier transform
- 2D Fourier transform
  - Definition
  - Examples and properties
- Convolution theorem
- Understanding the sampling theorem

# Fourier analysis

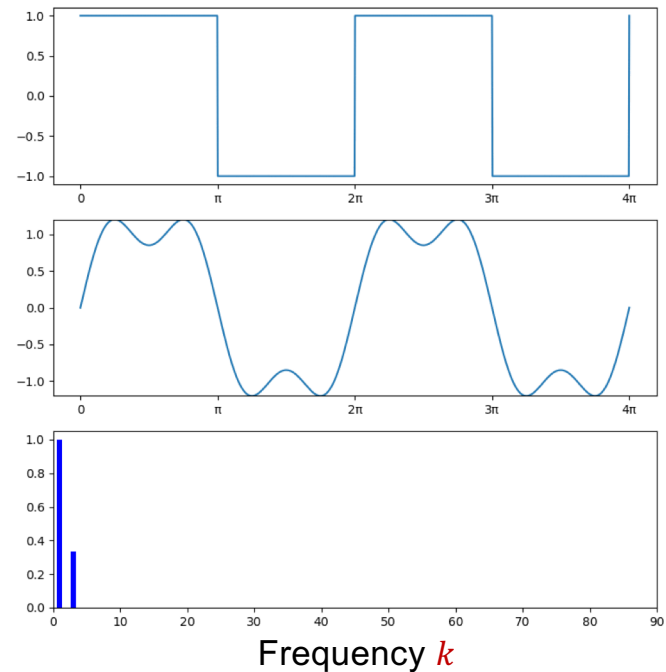
- Any(\*\*) univariate function can be expressed as a weighted sum of sinusoids of different frequencies (1807)



Jean-Baptiste Joseph Fourier (1768-1830)

Example: series for a square wave

$$\sum_{k=1,3,5,\dots}^{\infty} \frac{1}{k} \sin(kt)$$





# Fourier analysis

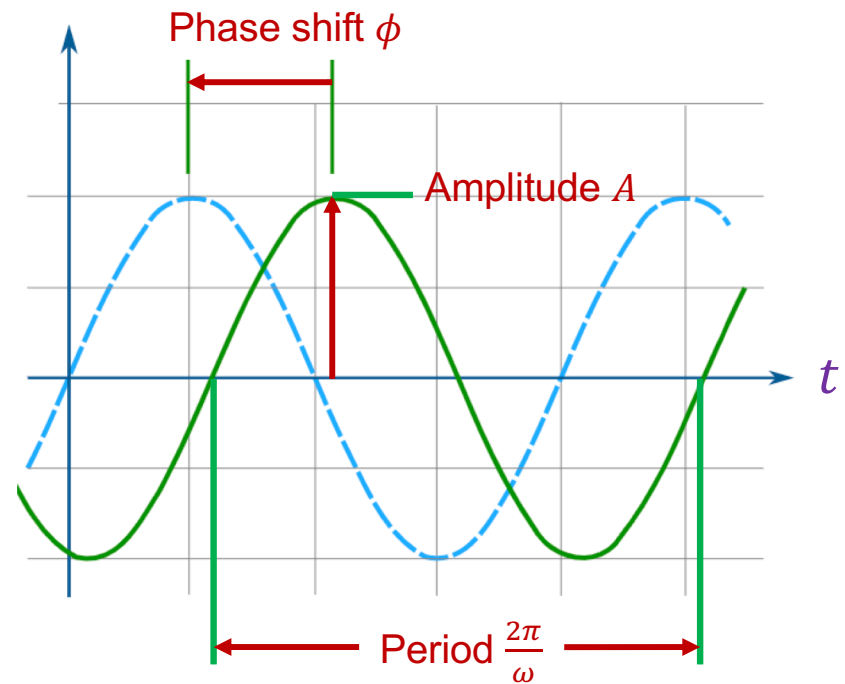
---

- Our building block:

$$A \sin(\omega t + \phi)$$

↑ Amplitude      ↓ Frequency      ↑ Phase

- Add enough of these to get any signal you want!



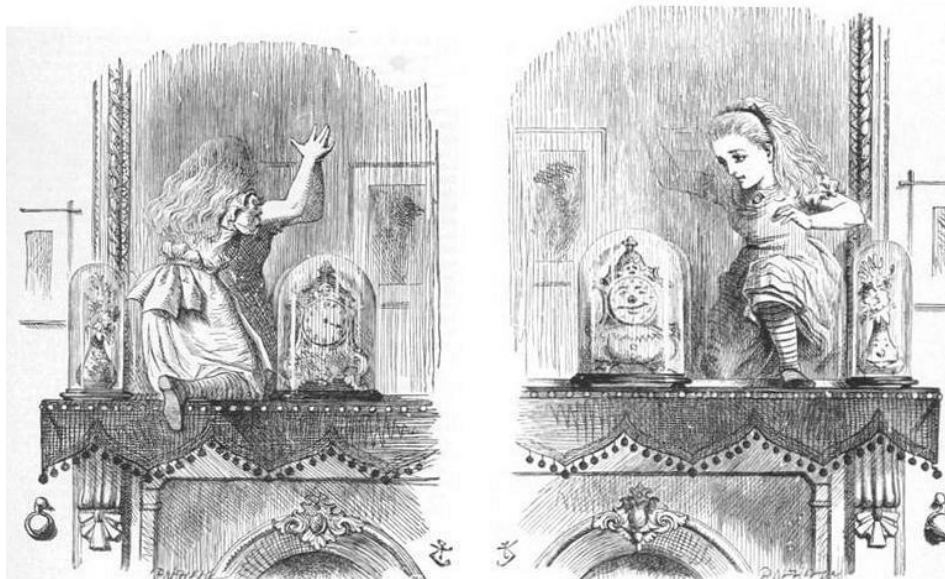
# 1D Fourier transform

---

- Let's define an (overcomplete) set of basis functions:

$$\psi_u(t) = e^{i2\pi ut}, \quad u \in (-\infty, \infty)$$

- Time to enter the strange world of complex exponentials...



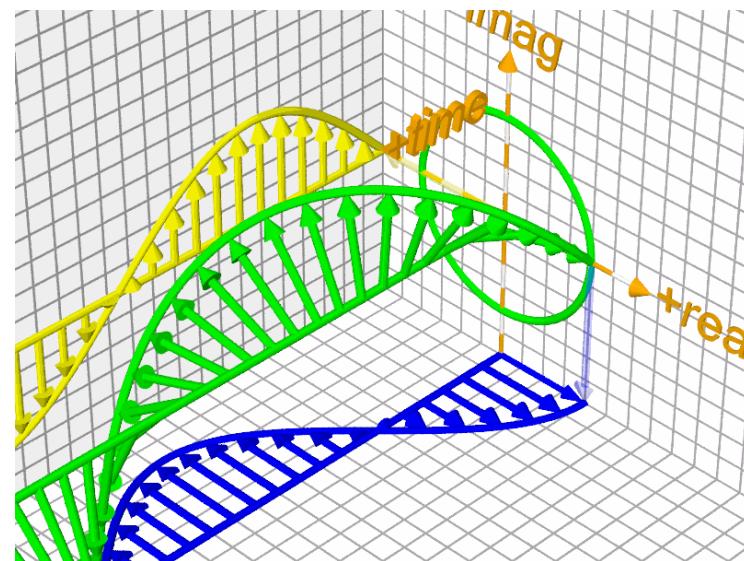
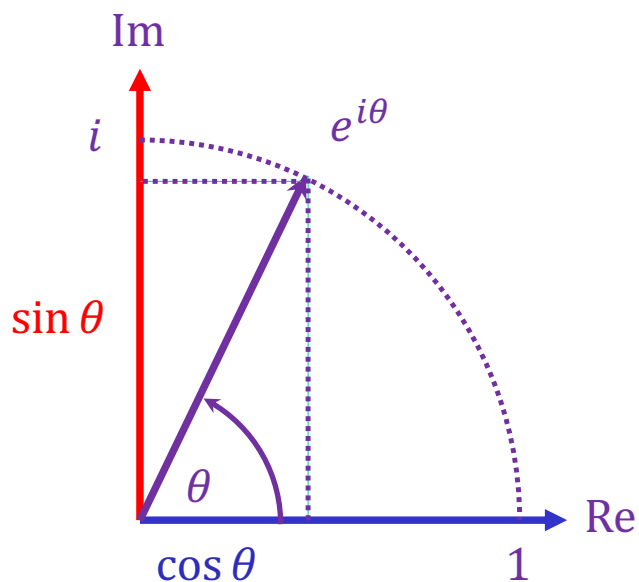
# 1D Fourier transform

---

- Let's define an (overcomplete) set of basis functions:

$$\psi_u(t) = e^{i2\pi ut}, \quad u \in (-\infty, \infty)$$

- Euler's formula:  $e^{i\theta} = \cos \theta + i \sin \theta$



[Image source](#)

# 1D Fourier transform

---

- Let's define a (continuously parameterized) set of basis functions:

$$\psi_u(t) = e^{i2\pi ut}, \quad u \in (-\infty, \infty)$$

- Euler's formula:  $e^{i\theta} = \cos \theta + i \sin \theta$
- So  $\psi_u(t)$  is just a cosine-sine pair at frequency  $u$ !

# 1D Fourier transform

---

- Let's define a (continuously parameterized) set of basis functions:

$$\psi_u(t) = e^{i2\pi ut}, \quad u \in (-\infty, \infty)$$

- Inner product for complex functions is given by:

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

Complex conjugate:  
real part stays the same,  
imaginary part is flipped

# 1D Fourier transform

---

- Let's define a (continuously parameterized) set of basis functions:

$$\psi_u(t) = e^{i2\pi ut}, \quad u \in (-\infty, \infty)$$

- Inner product for complex functions is given by:

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

- Orthonormality:

$$\langle \psi_{u_1}, \psi_{u_2} \rangle = \begin{cases} 1 & \text{if } u_1 = u_2 \\ 0 & \text{otherwise} \end{cases}$$



# 1D Fourier transform

---

- Given a signal  $f(t)$ , we want to represent it as a weighted combination of the basis functions  $\psi_u(t) = e^{i2\pi ut}$  with weights  $F(u)$ :

$$f(t) = \int_{-\infty}^{\infty} F(u) e^{i2\pi ut} du$$

- Each weight  $F(u)$  is given by the inner product of  $f$  and  $\psi_u$ :

$$F(u) = \langle f, \psi_u \rangle = \int_{-\infty}^{\infty} f(t) e^{-i2\pi ut} dt$$

# 1D Fourier transform

---

- Forward transform:  $f(t) \xrightarrow{\mathcal{F}} F(u)$

$$F(u) = \int_{-\infty}^{\infty} f(t) e^{-i2\pi ut} dt$$

- Note: for the FT to exist, the *energy*  $\int_{-\infty}^{\infty} |f(t)|^2 dt$  has to be finite

# 1D Fourier transform

---

- Forward transform:  $f(t) \xrightarrow{\mathcal{F}} F(u)$

$$F(u) = \int_{-\infty}^{\infty} f(t) e^{-i2\pi ut} dt$$

- For each  $u$ ,  $F(u)$  is a *complex number* that encodes both the *amplitude*  $A$  and *phase*  $\phi$  of the sinusoid  $A \sin(2\pi ut + \phi)$  in the decomposition of  $f(t)$ :

$$F(u) = \operatorname{Re}(F(u)) + i \operatorname{Im}(F(u)),$$

$$A = \sqrt{\operatorname{Re}(F(u))^2 + \operatorname{Im}(F(u))^2}, \quad \phi = \tan^{-1} \frac{\operatorname{Im}(F(u))}{\operatorname{Re}(F(u))}$$

- If  $f(t)$  is real, then  $\operatorname{Re}(F(u)) = \operatorname{Re}(F(-u))$ ,  
 $\operatorname{Im}(F(u)) = -\operatorname{Im}(F(-u))$

# 1D Fourier transform

---

- Forward transform:  $f(t) \xrightarrow{\mathcal{F}} F(u)$

$$F(u) = \int_{-\infty}^{\infty} f(t) e^{-i2\pi ut} dt$$

- Important properties:

- Energy preservation:

$$\int_{-\infty}^{\infty} |f(t)|^2 dt = \int_{-\infty}^{\infty} |F(u)|^2 du$$

- Linearity:  $\mathcal{F}\{af_1 + bf_2\} = a\mathcal{F}\{f_1\} + b\mathcal{F}\{f_2\}$

# 1D Fourier transform

---

- Forward transform:  $f(t) \xrightarrow{\mathcal{F}} F(u)$

$$F(u) = \int_{-\infty}^{\infty} f(t) e^{-i2\pi ut} dt$$

- Inverse transform:  $F(u) \xrightarrow{\mathcal{F}^{-1}} f(t)$

$$f(t) = \int_{-\infty}^{\infty} F(u) e^{i2\pi ut} du$$

- Duality: if  $f(t) \xrightarrow{\mathcal{F}} F(u)$ , then  $F(t) \xrightarrow{\mathcal{F}} f(-u)$ 
  - Thus, we can talk about *Fourier transform pairs*  $f(t) \leftrightarrow F(u)$

# Important Fourier transform pairs

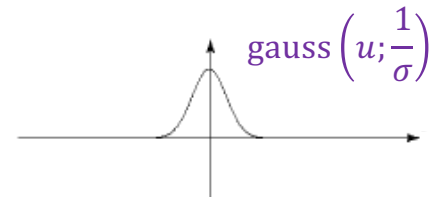
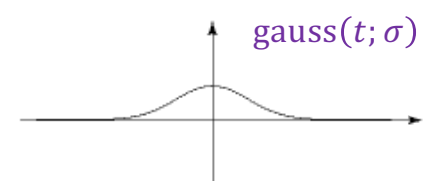
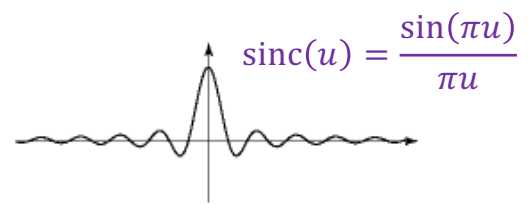
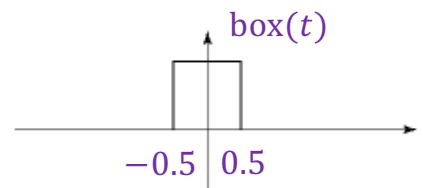
---





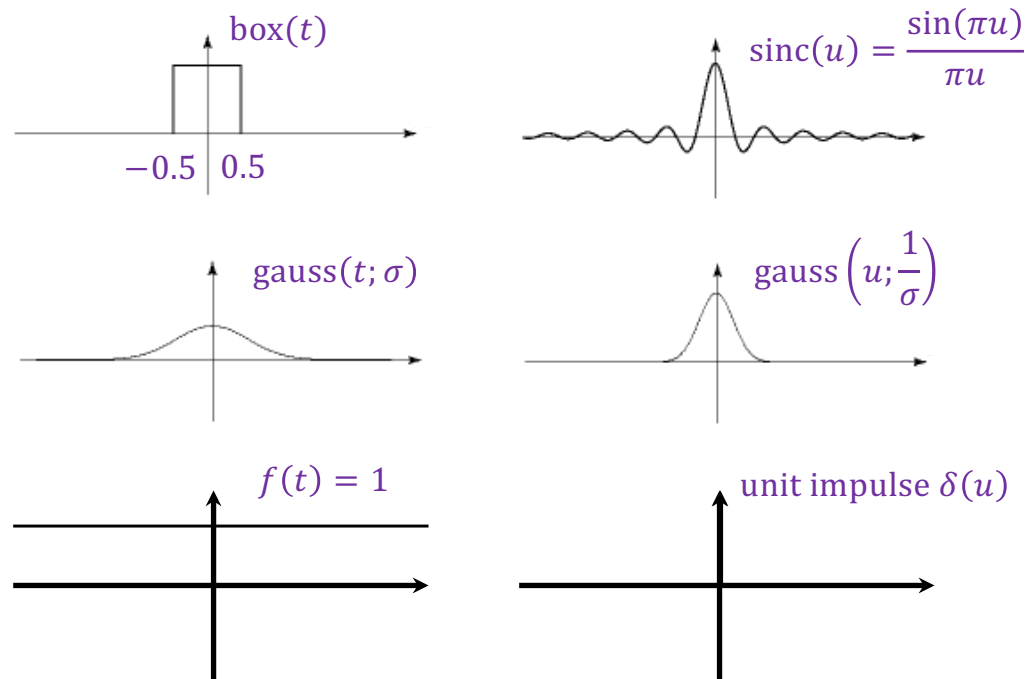
# Important Fourier transform pairs

---



# Important Fourier transform pairs

---



\*The last one is formal since these functions don't meet the mathematical requirements for FT

# Outline

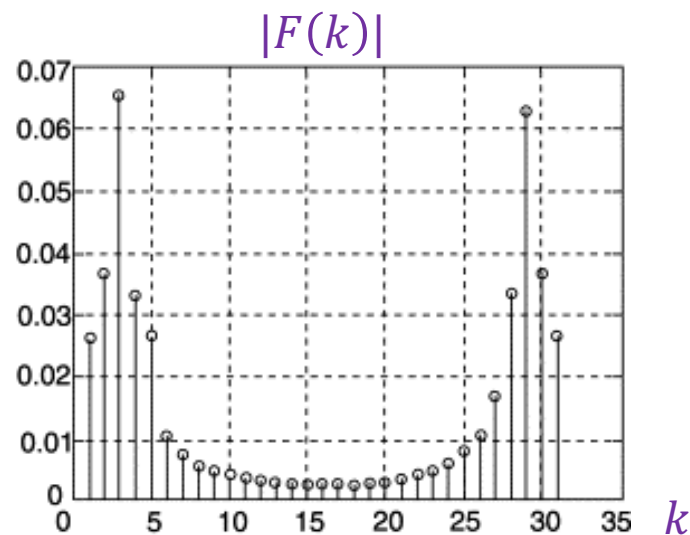
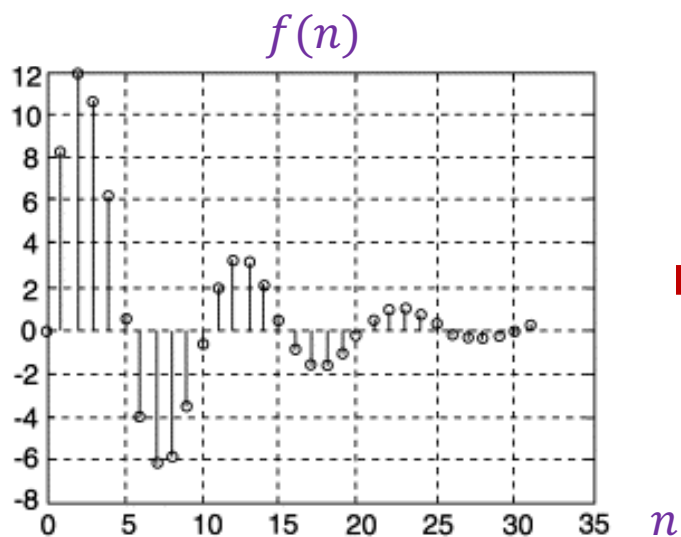
---

- 1D Fourier transform
  - Definition and properties
  - Discrete Fourier transform

# Discrete Fourier transform (DFT)

---

- Now suppose our signal consists of  $N$  samples  $f(n)$ ,  
 $n = 0, \dots, N - 1$
- We can also discretize frequencies to  $k/N$ ,  $k = 0, \dots, N - 1$   
( $k$  cycles per  $N$  samples)



[Image source](#)

## Discrete Fourier transform (DFT)

---

- Now suppose our signal consists of  $N$  samples  $f(n)$ ,  
 $n = 0, \dots, N - 1$
- We can also discretize frequencies to  $k/N$ ,  $k = 0, \dots, N - 1$   
( $k$  cycles per  $N$  samples)
- DFT formula:

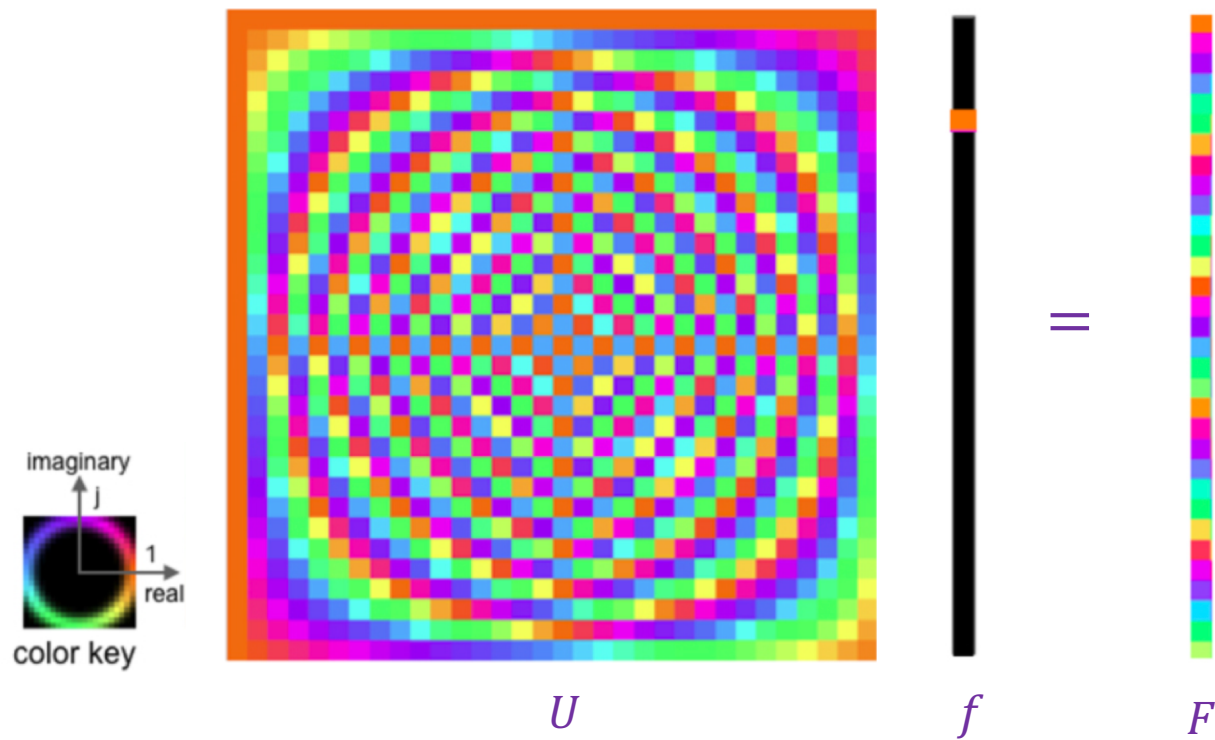
$$F(k) = \sum_{n=0}^{N-1} f(n) \exp\left(-i \frac{2\pi k}{N} n\right)$$

- We can pack the values  $\exp\left(-i \frac{2\pi k}{N} n\right)$ ,  $k, n = 0, \dots, N - 1$  into an  $N \times N$  matrix  $U$ , and DFT becomes just a matrix-vector multiplication!
- Fast Fourier transform: only  $N \log N$  complexity!

# DFT: Just a change of basis!

---

$$U f = F$$



[Source](#)



## Inverse DFT

---

- Forward DFT:

$$F(k) = \sum_{n=0}^{N-1} f(n) \exp\left(-i \frac{2\pi}{N} kn\right) \quad \text{or } F = Uf$$

- Inverse DFT:

$$f(n) = \frac{1}{N} \sum_{k=0}^{N-1} F(k) \exp\left(i \frac{2\pi}{N} kn\right) \quad \text{or } f = \frac{1}{N} U^{-1}F$$

where  $U^{-1}$  is the transpose of the *complex conjugate* of  $U$

## Periodicity of DFT and inverse DFT

---

- The result of DFT is periodic: because  $F(k)$  is obtained as a sum of complex exponentials with a common period of  $N$  samples,  $F(k + aN) = F(k)$  for any integer  $a$ :

$$\begin{aligned} F(k + aN) &= \sum_{n=0}^{N-1} f(n) \exp\left(-i \frac{2\pi}{N} n(k + aN)\right) \\ &= \sum_{n=0}^{N-1} f(n) \exp\left(-i \frac{2\pi n}{N} k\right) \exp(-i2\pi an) = F(k) \end{aligned}$$

- Likewise, the result of the inverse DFT is a periodic signal:  $f(t + aN) = f(t)$  for any integer  $a$

# Outline

---

- 1D Fourier transform
  - Definition and properties
  - Discrete Fourier transform
- 2D Fourier transform

## 2D Fourier transform

---

- To represent 2D signals  $f(x, y)$ , we need to extend our 1D basis functions  $\psi_u(t) = e^{i2\pi ut}$  to two variables:

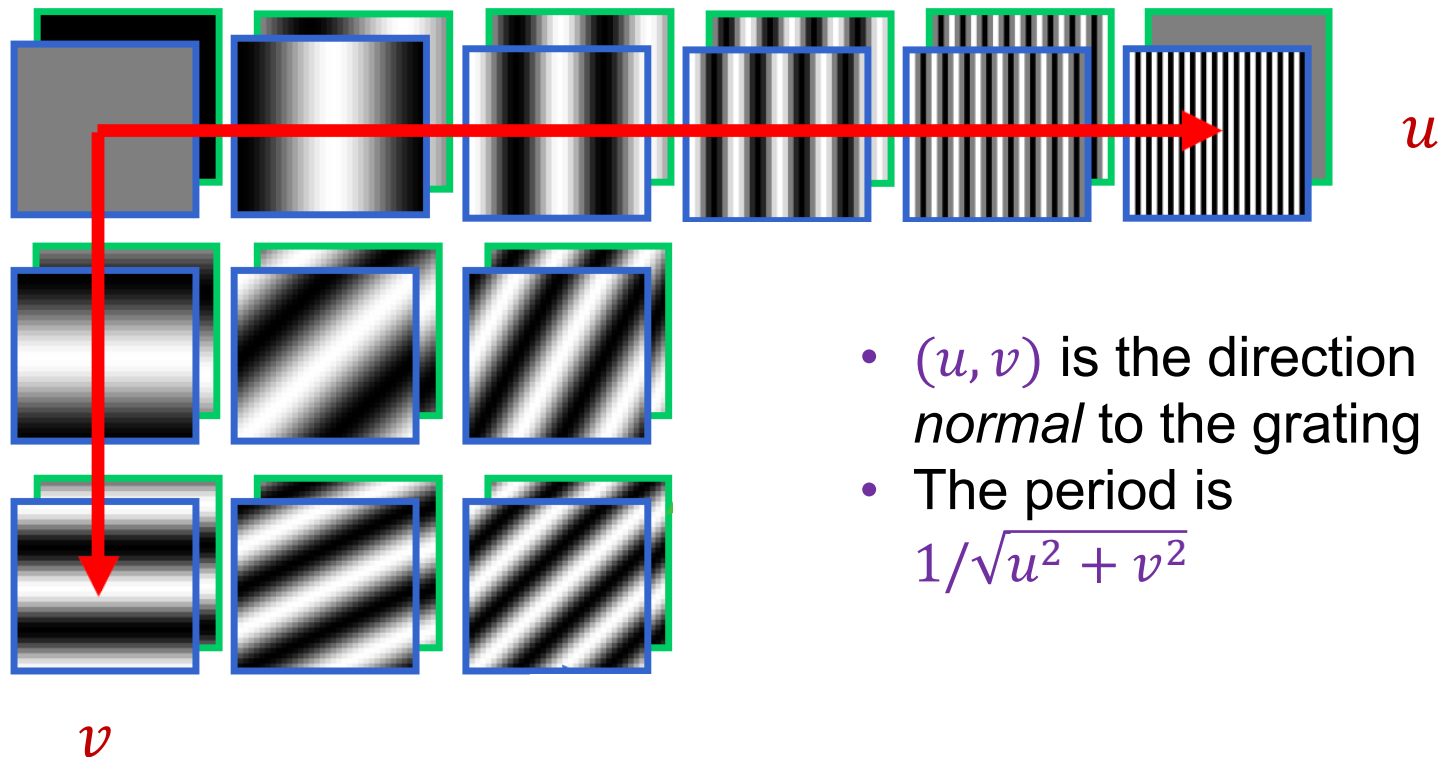
$$\begin{aligned}\psi_{u,v}(x, y) &= e^{i2\pi ux} e^{i2\pi vy} = e^{i2\pi(ux+vy)} \\ &= \cos 2\pi(ux + vy) + i \sin 2\pi(ux + vy)\end{aligned}$$

- What does this look like?

## 2D Fourier transform

---

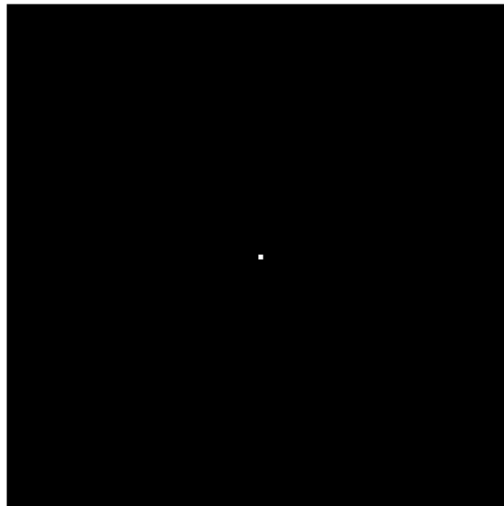
- 2D basis functions are oriented sinusoidal “gratings”:



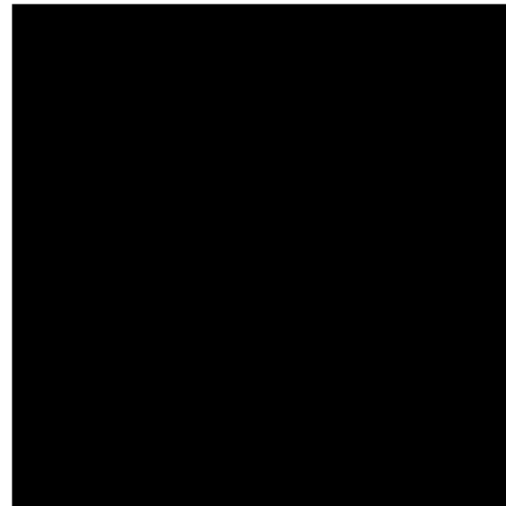
# Basis function examples

---

$(u, v)$



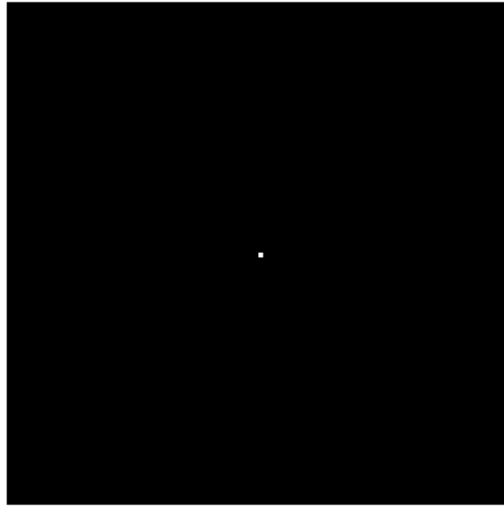
Real  
component



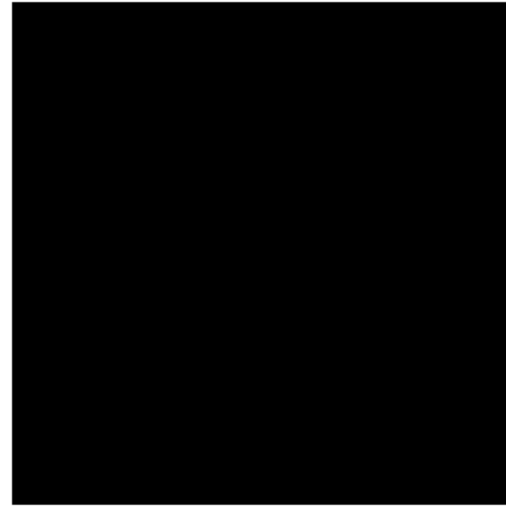
# Basis function examples

---

$(u, v)$



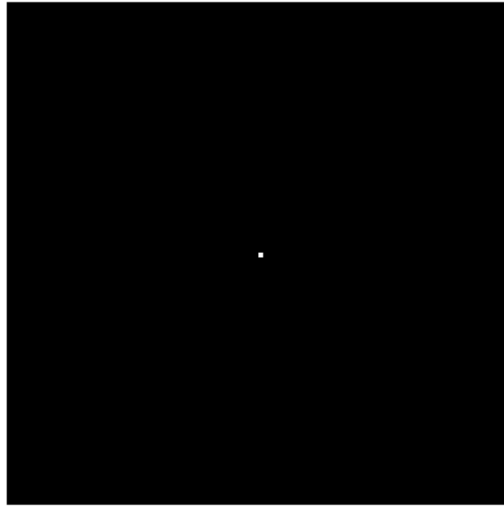
Real  
component



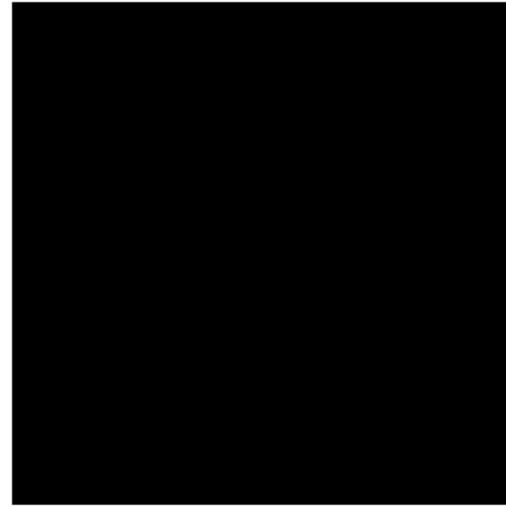
# Basis function examples

---

$(u, v)$



Real  
component

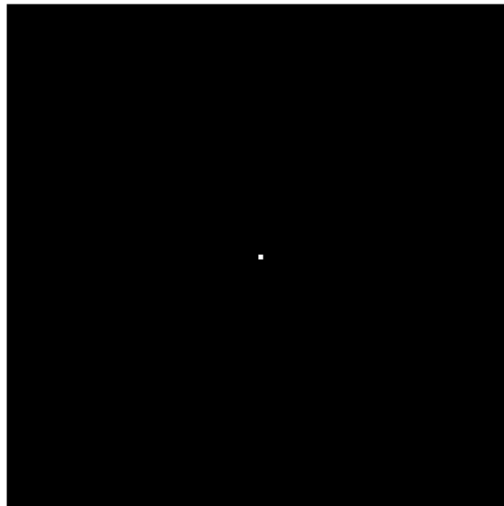




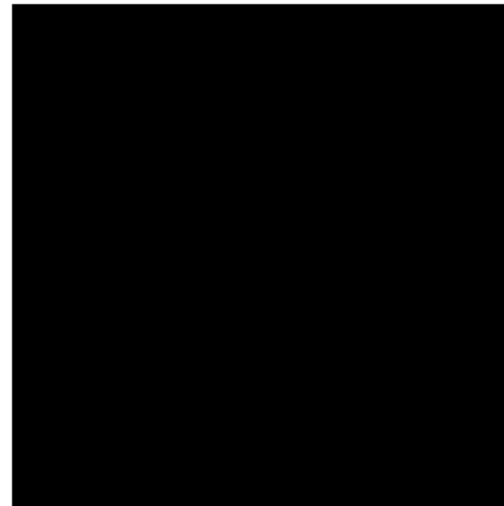
# Linear combination of basis functions

---

$(u, v)$



Real  
component



## 2D Fourier transform

---

$$F(u, v) = \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} f(x, y) e^{-i2\pi(ux+vy)} dx dy$$

- Output is 2D and complex-valued:

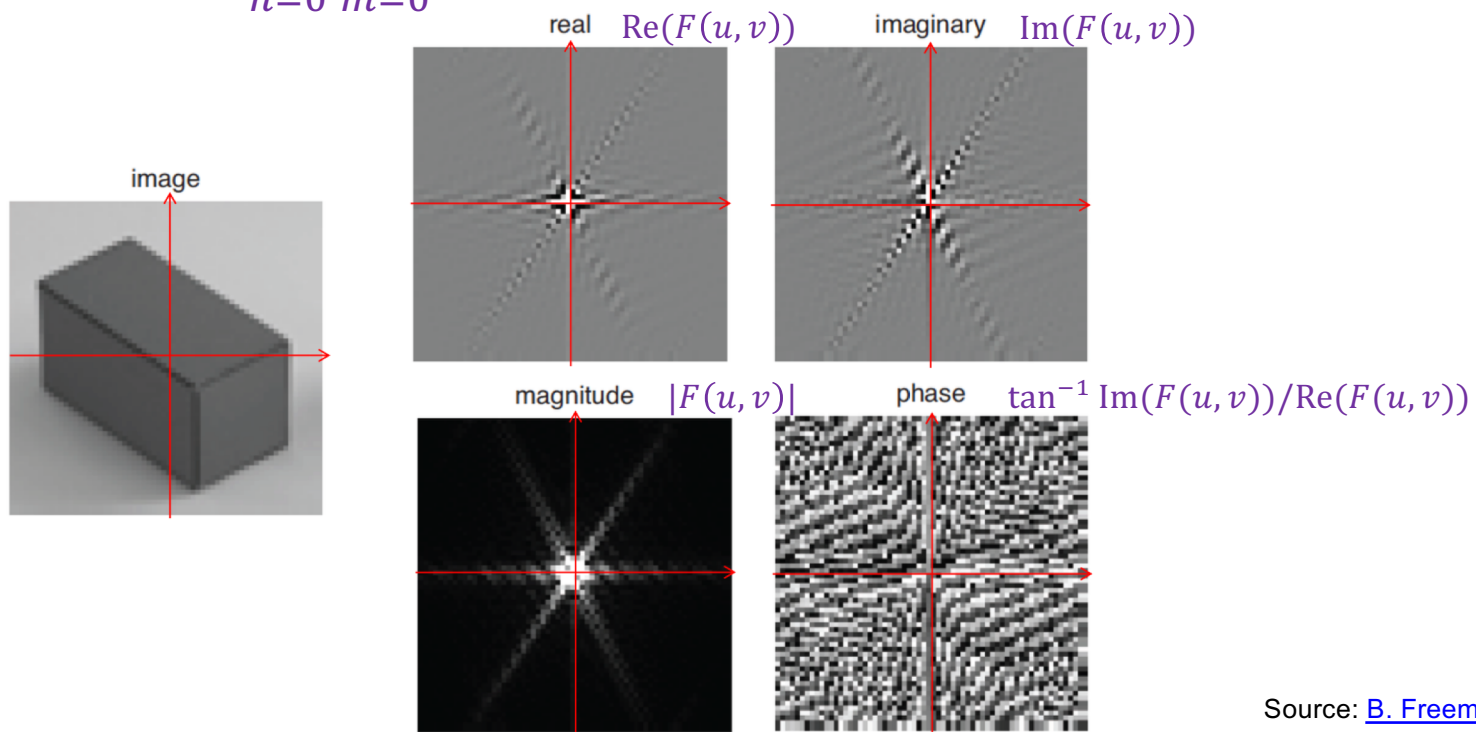
$$F(u, v) = \text{Re}(F(u, v)) + i \text{Im}(F(u, v))$$

- Magnitude spectrum:  $|F(u, v)| = \sqrt{\text{Re}(F(u, v))^2 + \text{Im}(F(u, v))^2}$
- Phase angle spectrum:  $\tan^{-1} \frac{\text{Im}(F(u, v))}{\text{Re}(F(u, v))}$
- Symmetry: the Fourier transform of a real-valued image has coefficients that come in pairs, with  $F(u, v)$  being the *complex conjugate* of  $F(-u, -v)$ 
  - This means that the magnitude spectrum is symmetric about the origin

# 2D discrete Fourier transform

---

$$F(u, v) = \sum_{n=0}^{N-1} \sum_{m=0}^{M-1} f(n, m) \exp\left(-i2\pi\left(\frac{un}{N} + \frac{vm}{M}\right)\right)$$



Source: [B. Freeman](#)

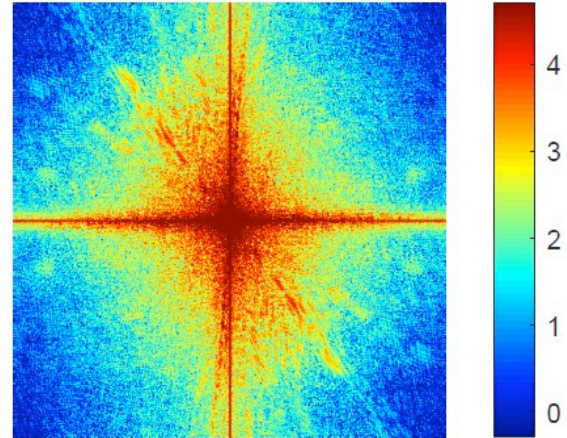
# Real image examples

---

intensity image



log fft magnitude



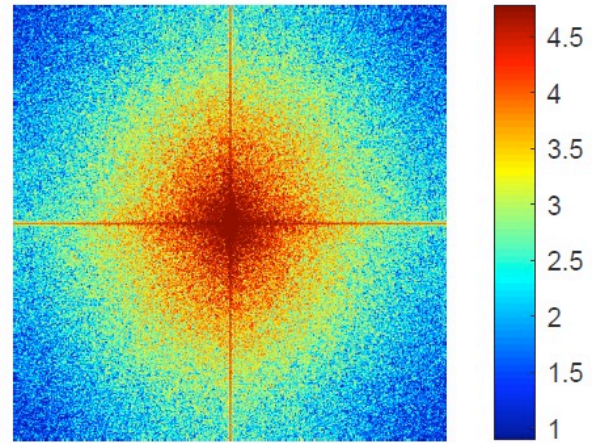
# Real image examples

---

intensity image

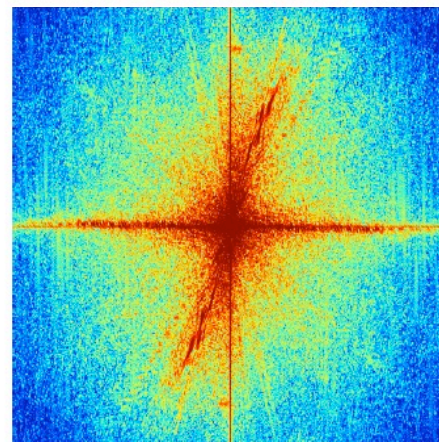
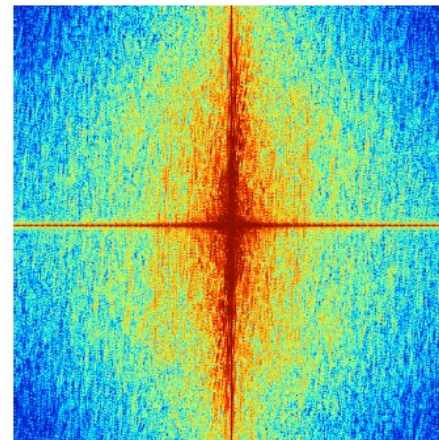


log fft magnitude



Which image goes with which spectrum?

---



## Phase vs. magnitude

---

- Which has more information, the phase or the magnitude?
- Let's take the phase from one image and combine it with the magnitude from another image

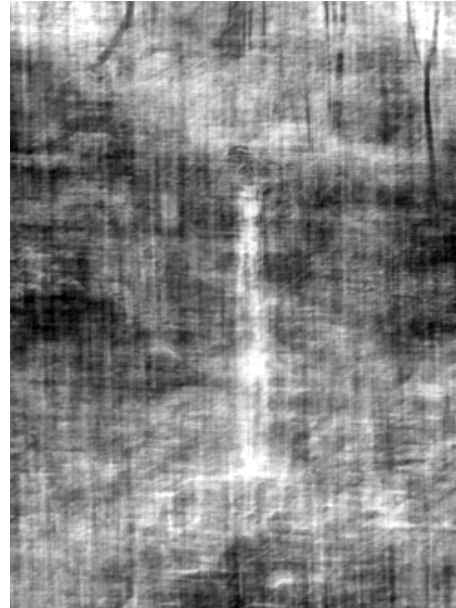




Magnitude



Phase







Phase



Magnitude

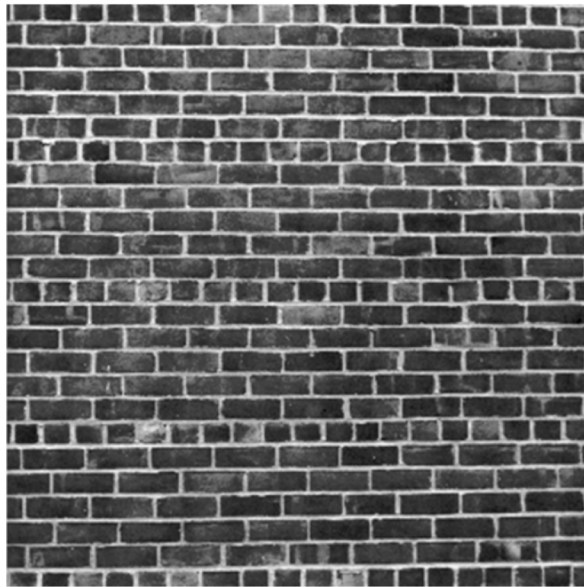


## Images with periodic patterns

---

- The magnitude image has peaks corresponding to the frequencies of repetition

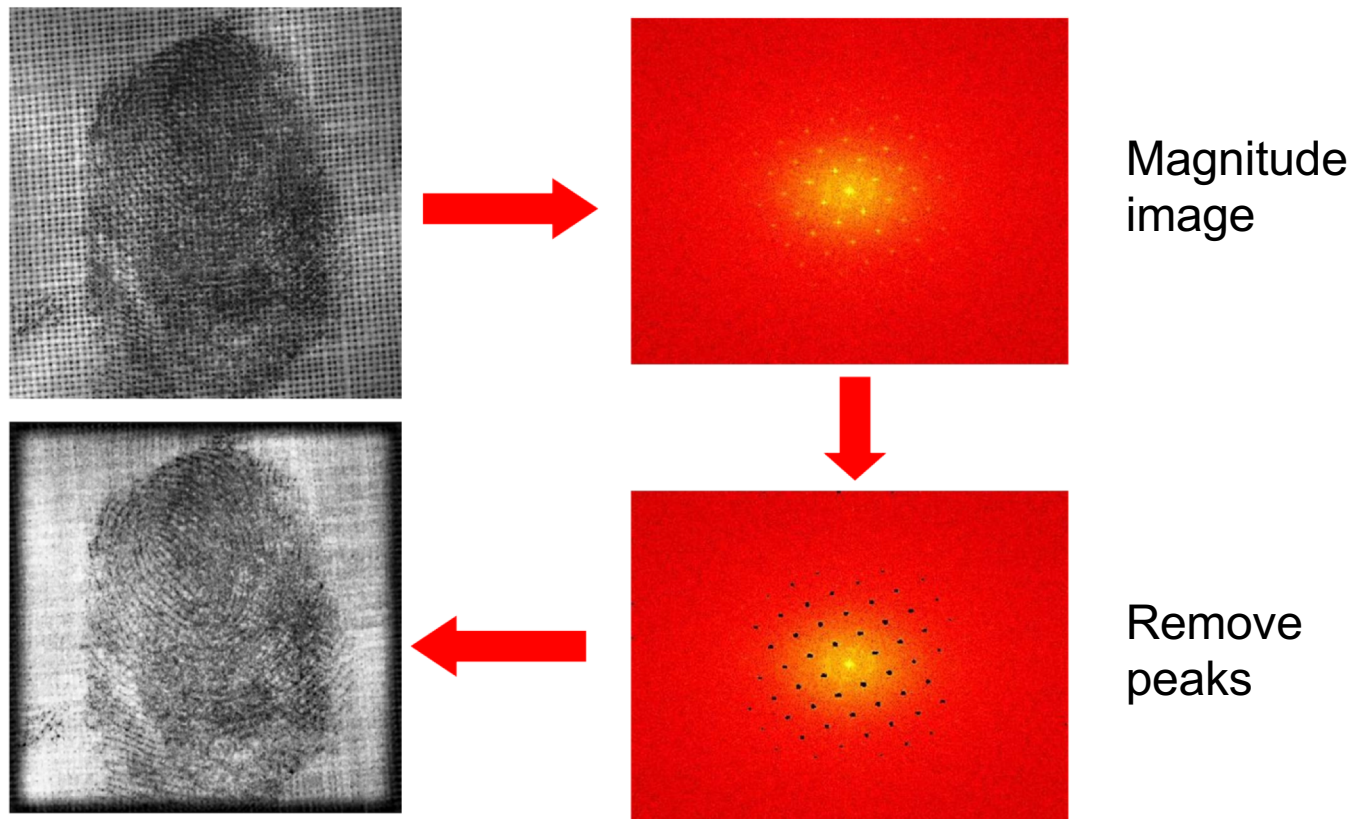
Image



Source: [A. Zisserman](#)

# Application: Removing periodic patterns

---



Source: [A. Zisserman](#)

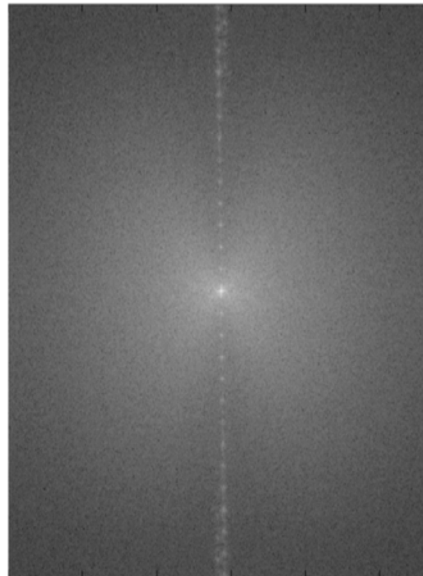
# Application: Removing periodic patterns

---

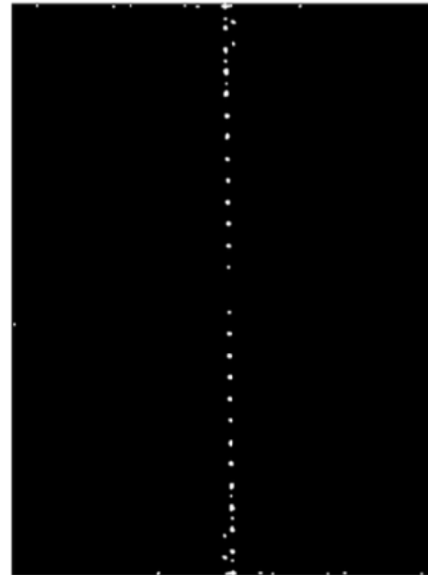
Lunar orbital image  
(1966)



Magnitude  
image



Remove  
peaks



Join lines  
removed

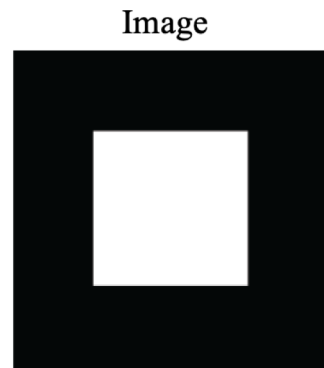
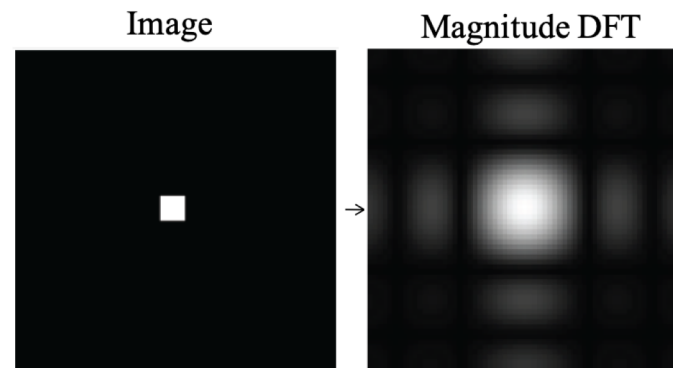


Source: [A. Zisserman](#)

# Image transformations

---

- How does the FT change when the image is scaled?

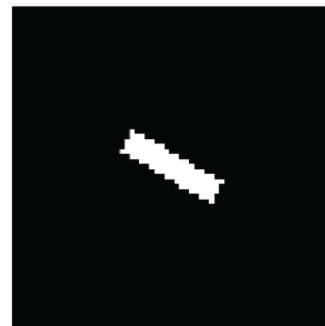
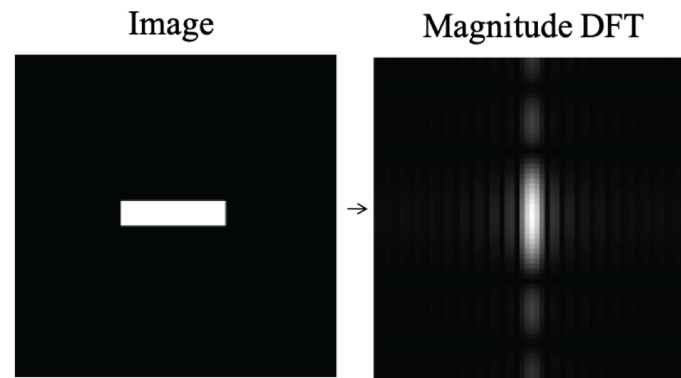


Scaled by the  
inverse factor!

# Image transformations

---

- How does the FT change when the image is rotated?

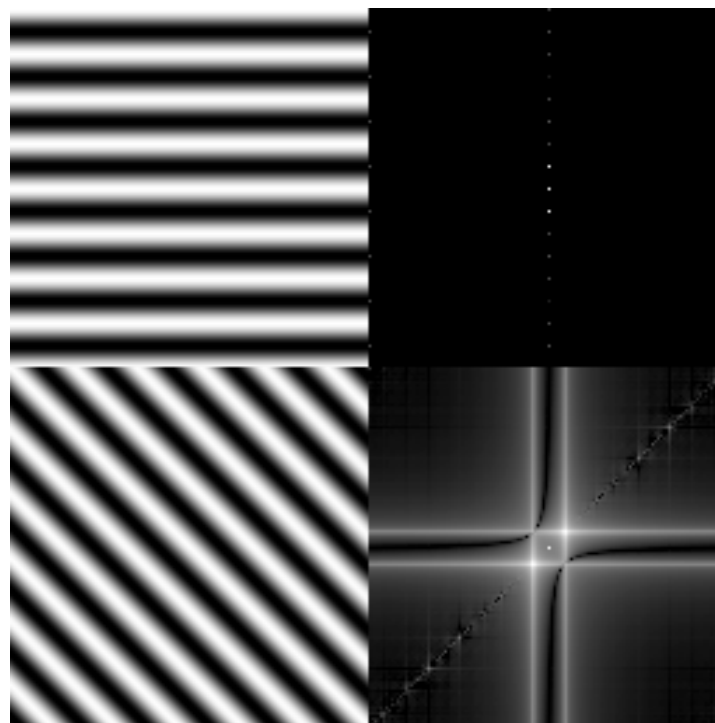


Rotates the same way!

# Image transformations

---

- How does the FT change when the image is rotated?

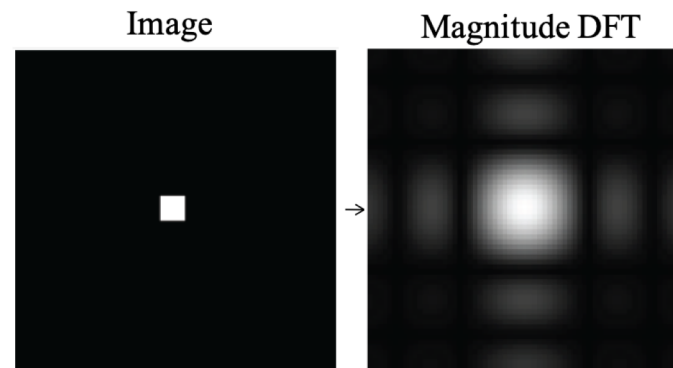


Caution: in real images this is not always the case because of edge artifacts (recall that DFT treats images as periodically tiled)

# Image transformations

---

- How does the FT change when the image is translated?



Magnitude spectrum  
doesn't change,  
phase gets  
modulated



# Outline

---

- 1D Fourier transform
  - Definition and properties
  - Discrete Fourier transform
- 2D Fourier transform
  - Definition
  - Examples and properties
- Convolution theorem

## Convolution theorem

---

- **Convolution** in the spatial domain translates to **multiplication** in the frequency domain (and vice versa)
- The Fourier transform of the convolution of two functions is the product of their Fourier transforms:

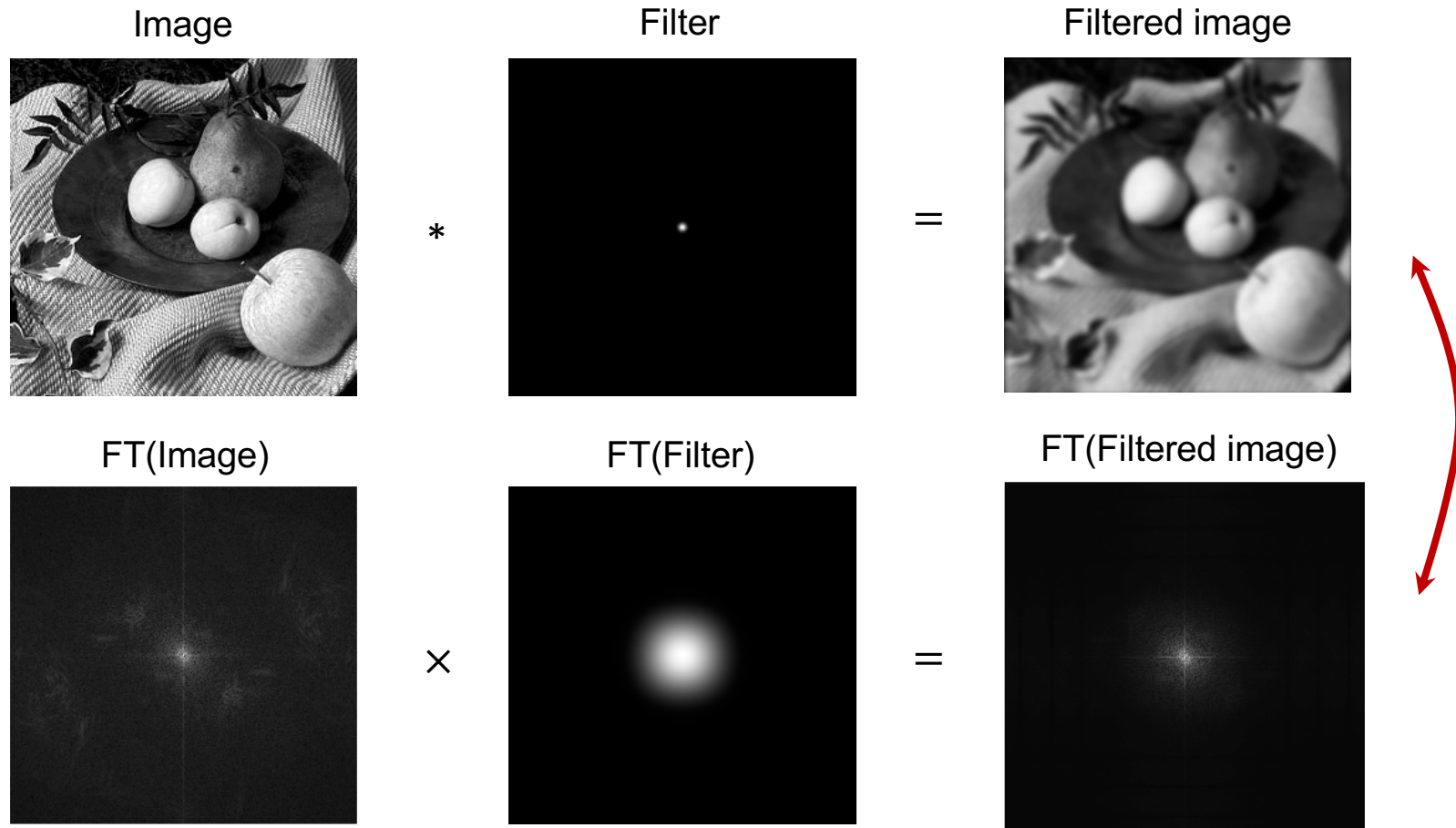
$$\mathcal{F}\{f * g\} = \mathcal{F}\{f\} \mathcal{F}\{g\}$$

- The inverse Fourier transform of the product of two Fourier transforms is the convolution of the two inverse Fourier transforms:

$$\mathcal{F}^{-1}\{FG\} = \mathcal{F}^{-1}\{F\} * \mathcal{F}^{-1}\{G\}$$

# 2D convolution theorem example

---



## Convolution theorem

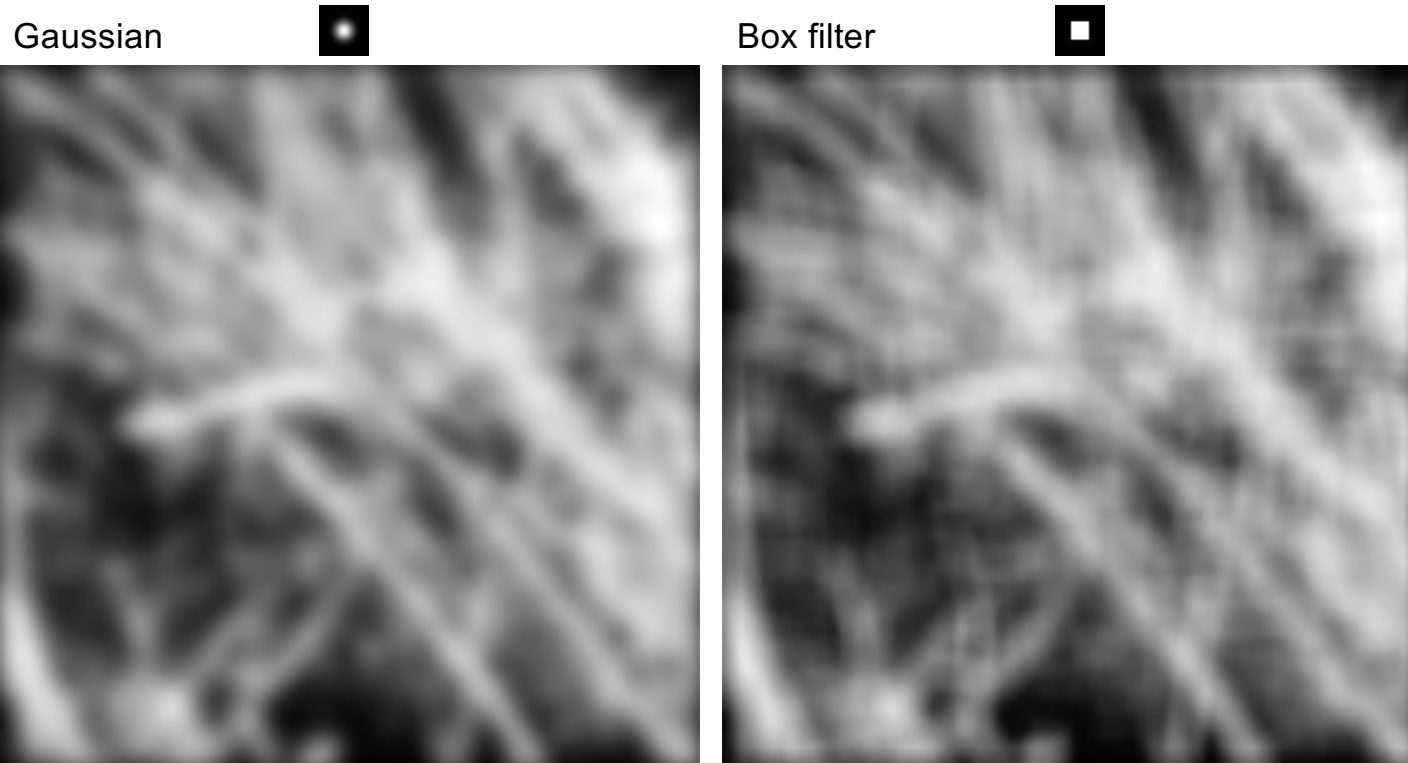
---

- Suppose  $f$  and  $g$  both consist of  $N$  pixels
- What is the complexity of computing  $f * g$  in the spatial domain?
  - $O(N^2)$
- And what is the complexity of computing  $\mathcal{F}^{-1}\{\mathcal{F}\{f\}\mathcal{F}\{g\}\}$ ?
  - $O(N \log N)$  using FFT
- Thus, convolution of an image with a large filter can be more efficiently done in the frequency domain

## Understanding the behavior of filtering

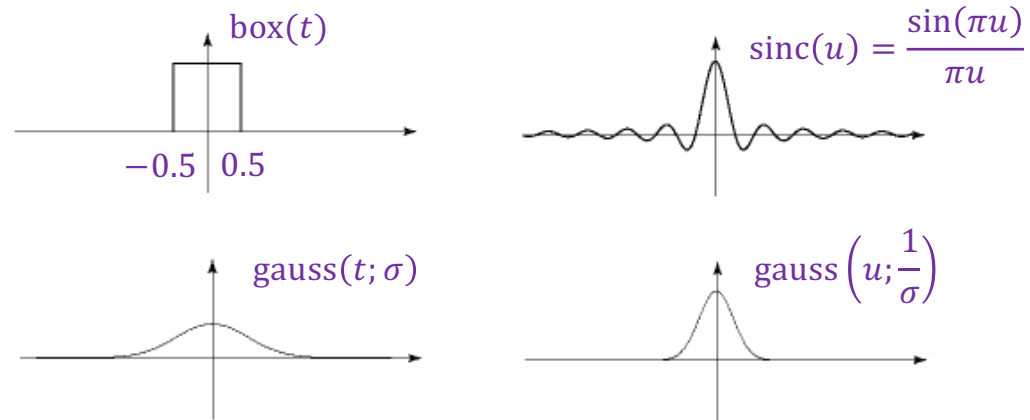
---

- Why does filtering with a Gaussian give a nice smooth image, but filtering with a box filter gives artifacts?



# Recall: Fourier transform pairs

---



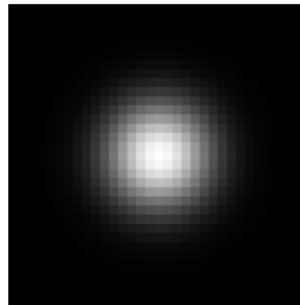
# Filtering with a Gaussian

---

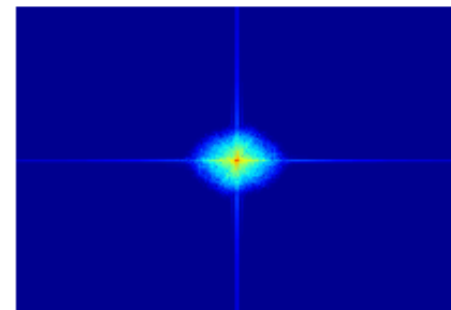
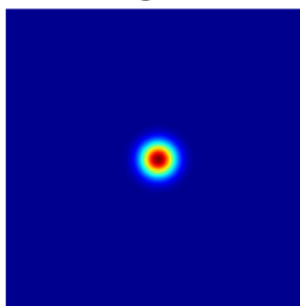
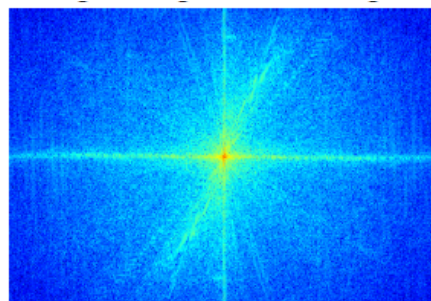
intensity image



filter: gaussian



filtered image



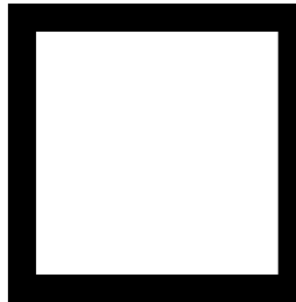
# Filtering with a box filter

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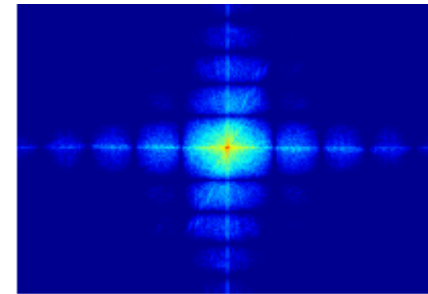
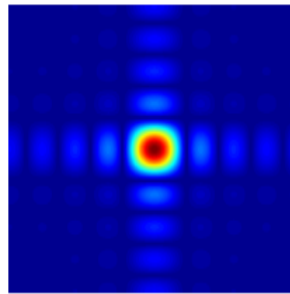
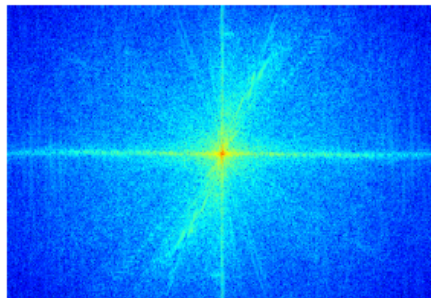
intensity image



filter: box



filtered image





# Low-pass and high-pass filtering

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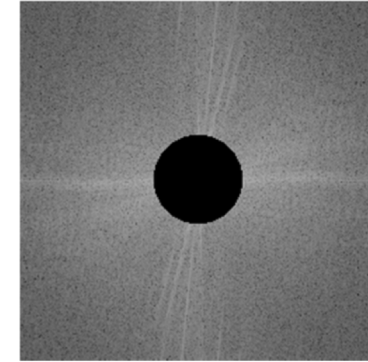
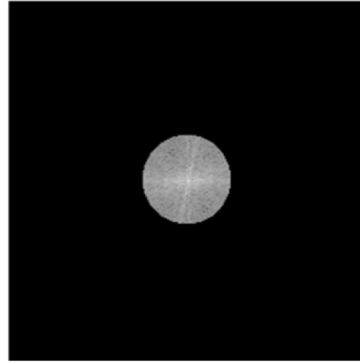
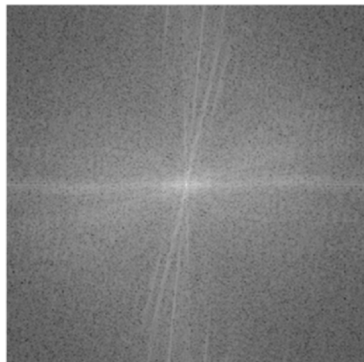
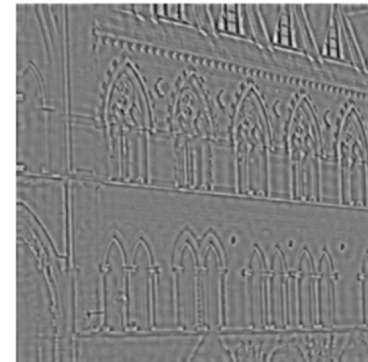
Image



Low-pass filtered



High-pass filtered



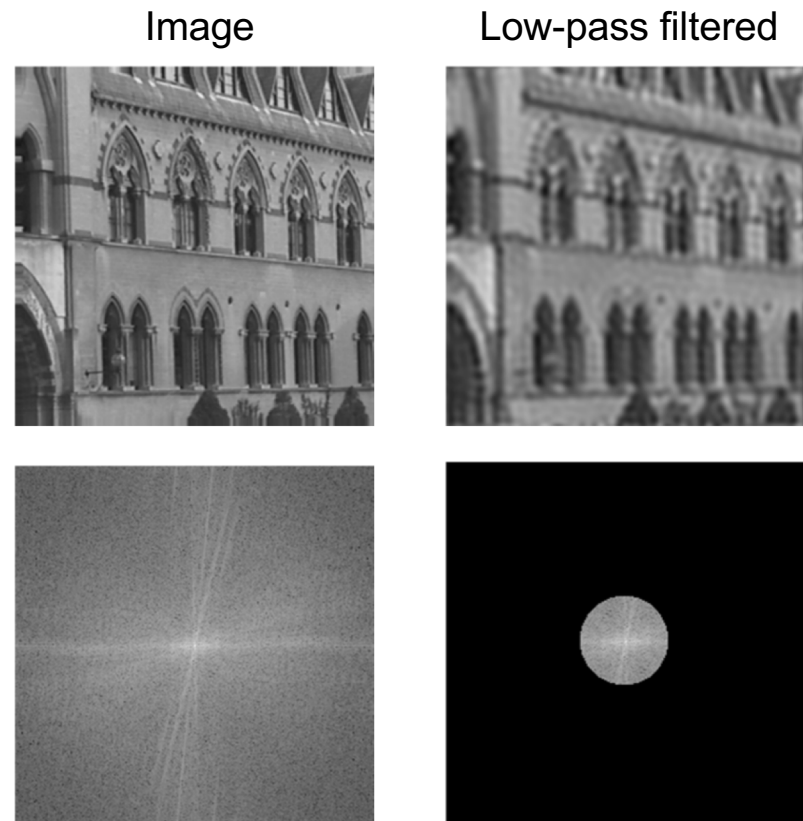
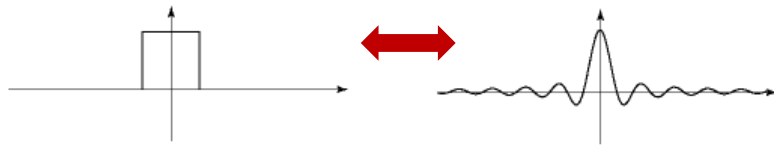
[Demo](#)

Source: [A. Zisserman](#)

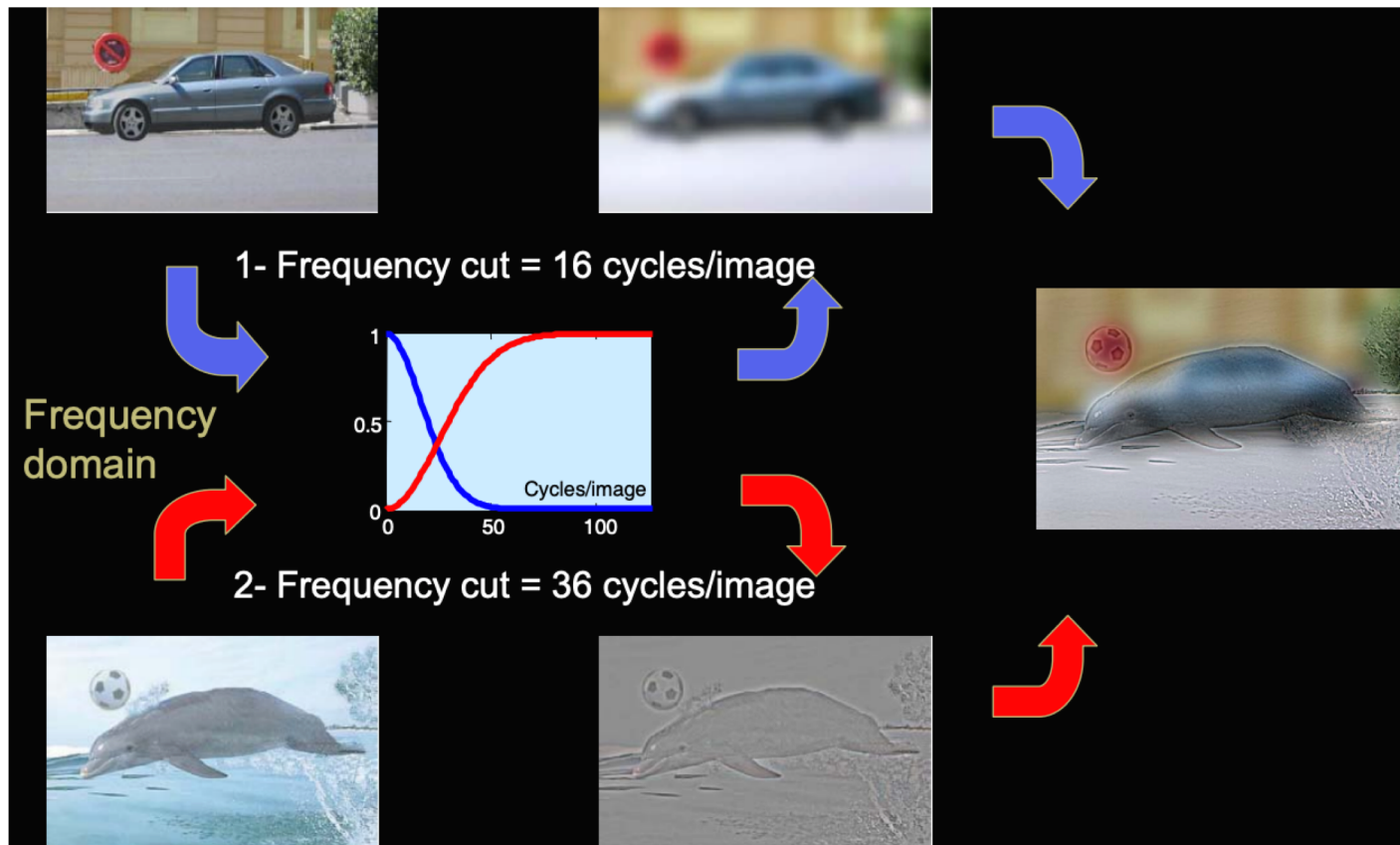
# Closer look at low-pass filtering

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- Do we like this low-pass filtering result?
- No – it causes ringing artifacts in the image (why?)
  - Recall: it's equivalent to convolving with a sinc function in the spatial domain
- This is why Gaussian filtering is preferred



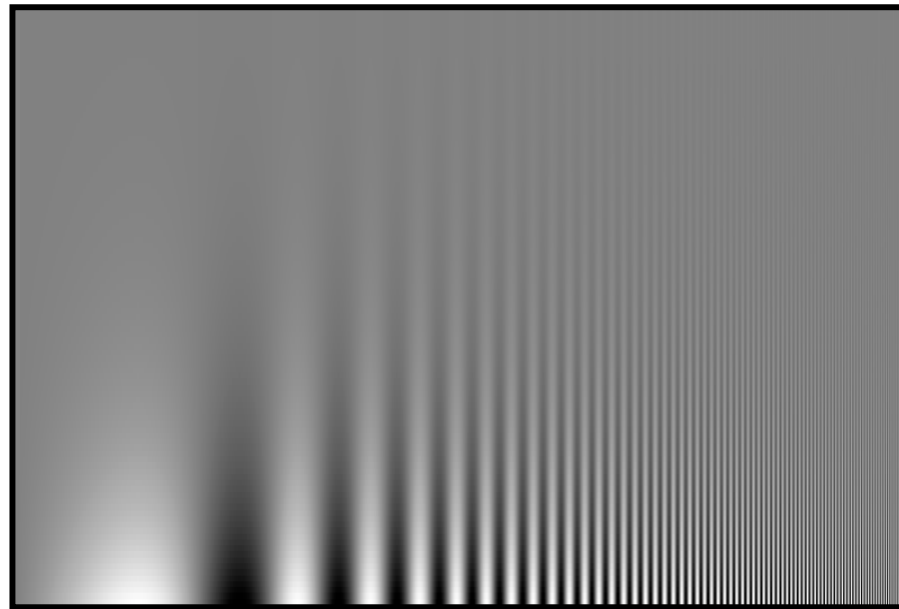
# Hybrid images in the frequency domain



## Human contrast sensitivity curve

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- Depending on viewing distance, peak sensitivity will occur at different frequencies



# Outline

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- 1D Fourier transform
  - Definition and properties
  - Discrete Fourier transform
- 2D Fourier transform
  - Definition
  - Examples and properties
- Convolution theorem
- Understanding the sampling theorem

# Understanding sampling and aliasing

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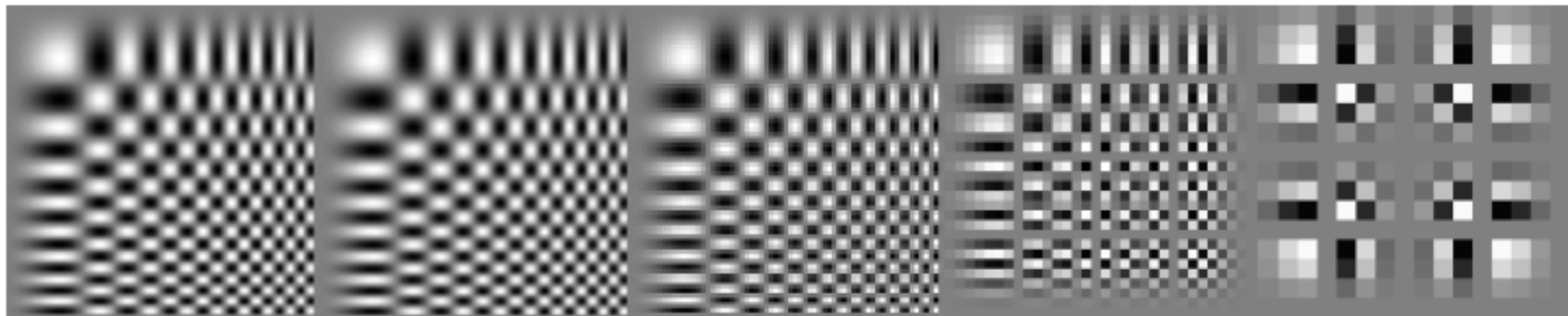
256x256

128x128

64x64

32x32

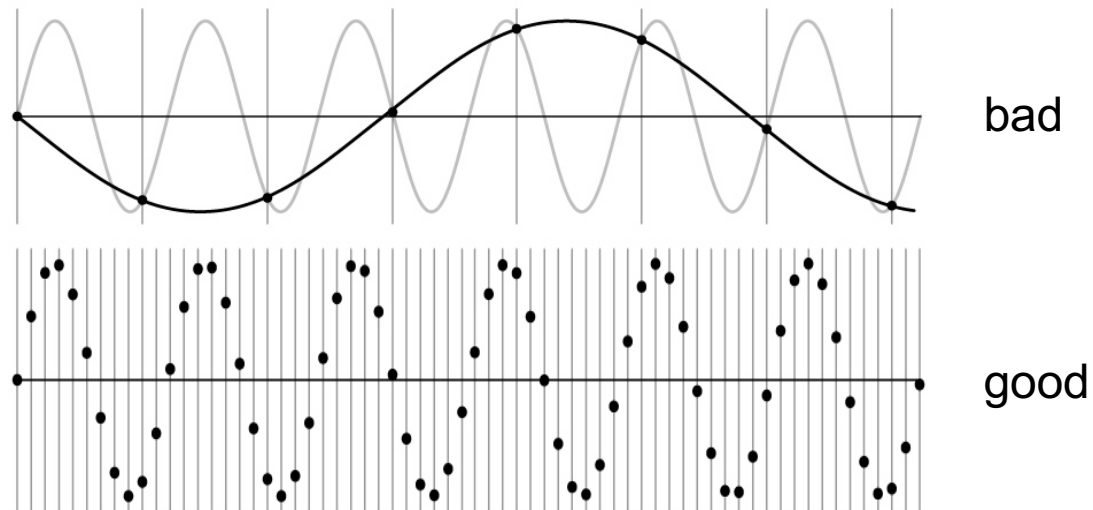
16x16



## Recall: Nyquist-Shannon sampling theorem

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- When sampling a signal at discrete intervals, the sampling frequency must be at least *twice* the maximum frequency of the input signal to allow us to reconstruct the original perfectly from the sampled version



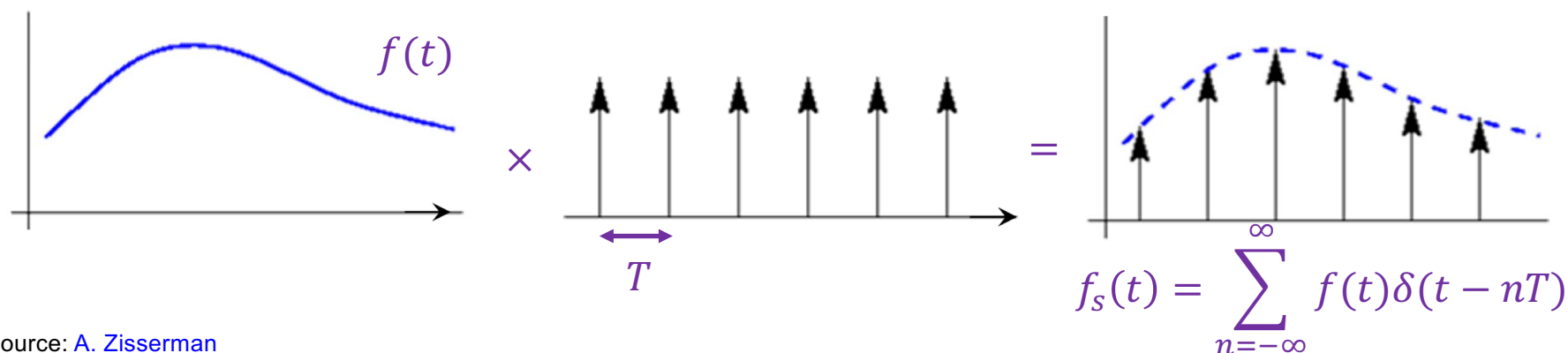
[https://en.wikipedia.org/wiki/Nyquist-Shannon\\_sampling\\_theorem](https://en.wikipedia.org/wiki/Nyquist-Shannon_sampling_theorem)

# Understanding the sampling theorem

---

- Suppose we have a continuous function  $f(t)$  and we want to sample it at discrete intervals with a spacing of  $T$
- This can be accomplished by multiplying it by the *comb function* or *impulse train*:

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

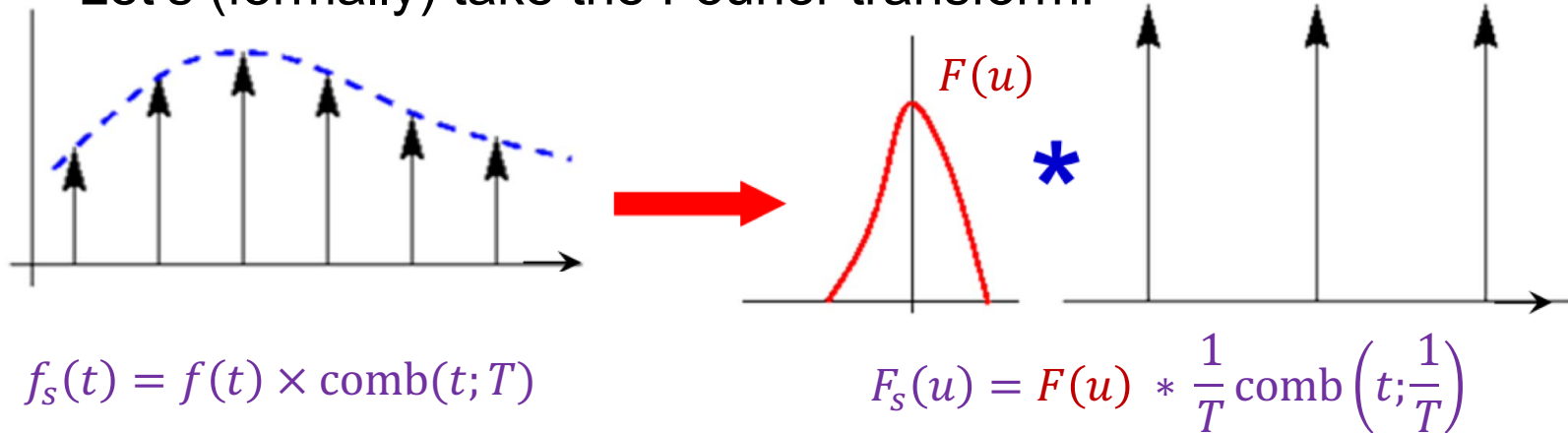




# Understanding the sampling theorem

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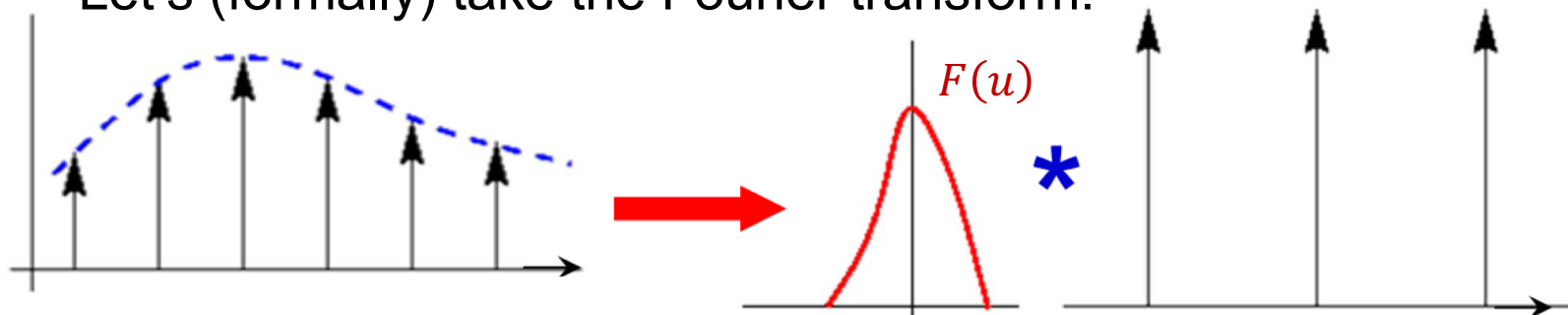
- Let's (formally) take the Fourier transform:



\*Officially, the FT of the comb function doesn't exist since it's periodic, and since  $\delta$  is a weird function

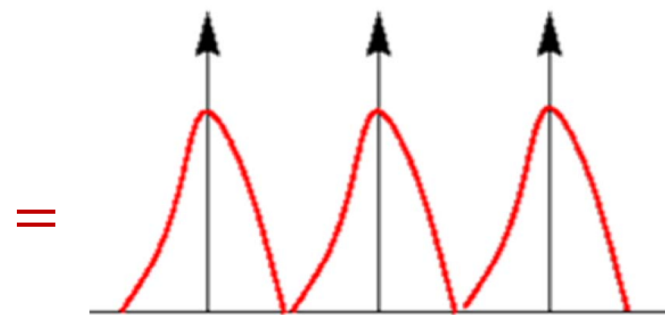
# Understanding the sampling theorem

- Let's (formally) take the Fourier transform:



$$f_s(t) = f(t) \times \text{comb}(t; T)$$

$$F_s(u) = F(u) * \frac{1}{T} \text{comb}\left(t; \frac{1}{T}\right)$$

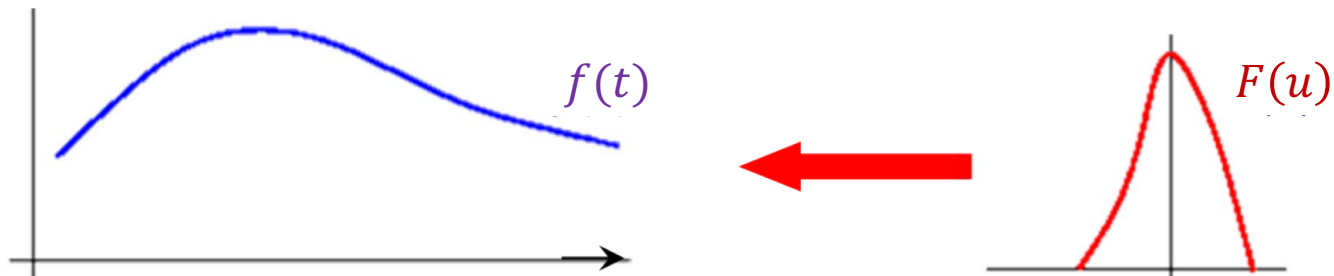
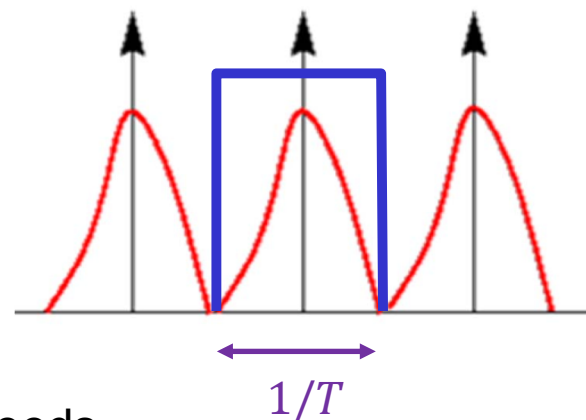


Replicated copies of  $F(u)$ !

# Understanding the sampling theorem

---

- How do we reconstruct  $f(t)$ ?
- Let's apply a box filter in the frequency domain (equivalent to convolving with a sinc function in the original domain)
- When will this succeed?
  - When the sampling frequency  $1/T$  exceeds twice the greatest frequency contained in  $F(u)$ !

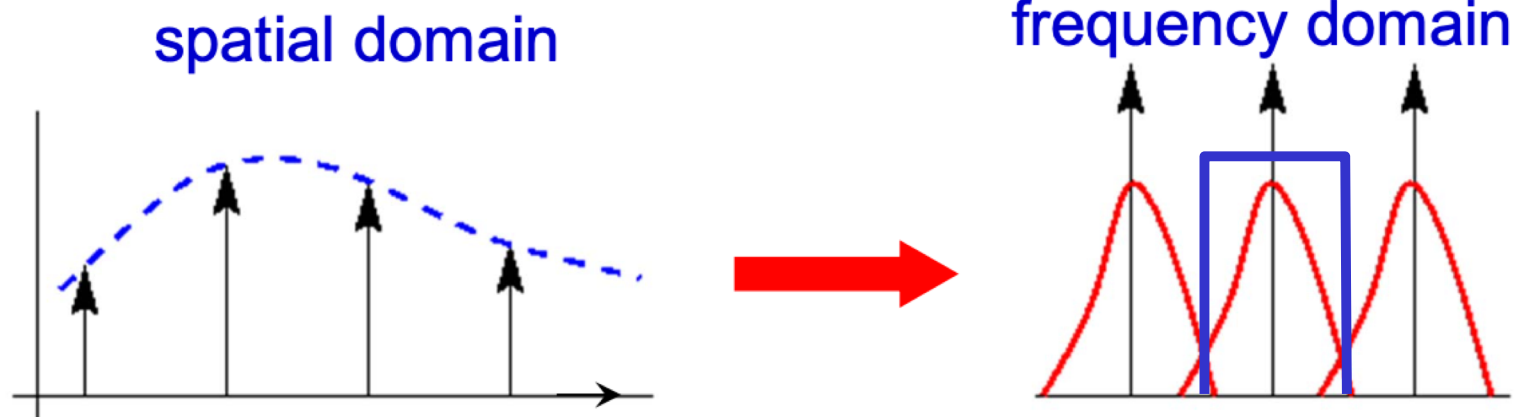


Source: [A. Zisserman](#)

# Understanding the sampling theorem

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- If the sampling frequency is too small, frequencies above the Nyquist limit are “folded back” onto smaller frequencies, resulting in aliasing

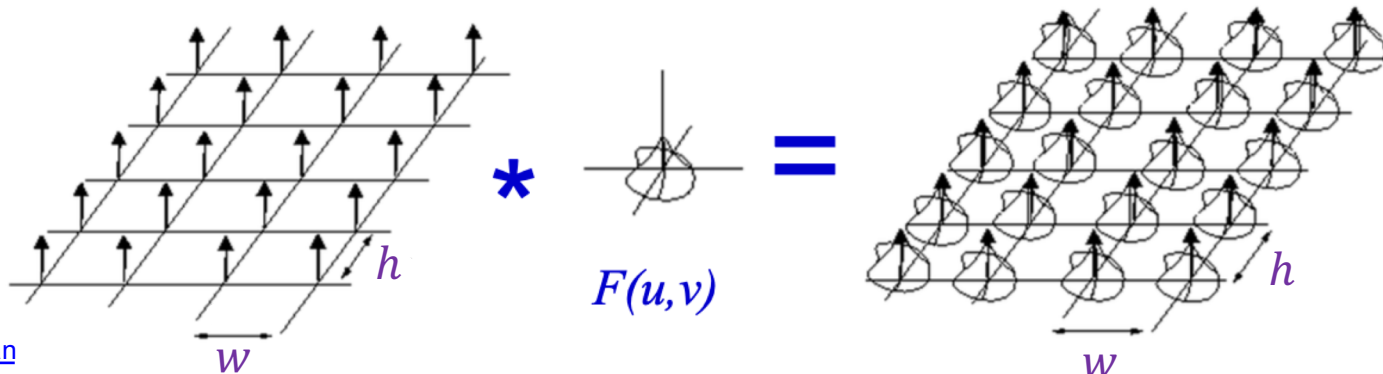


# Sampling theorem in 2D

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- If the Fourier transform of a continuous function  $f(x, y)$  is zero for all frequencies beyond  $u_b$  and  $v_b$ , i.e., if the Fourier transform is *band-limited*, then  $f(x, y)$  can be completely reconstructed from its samples as long as the sampling distances  $w$  and  $h$  along the  $x$  and  $y$  directions are such that  $w \leq \frac{1}{2u_b}$  and  $h \leq \frac{1}{2v_b}$

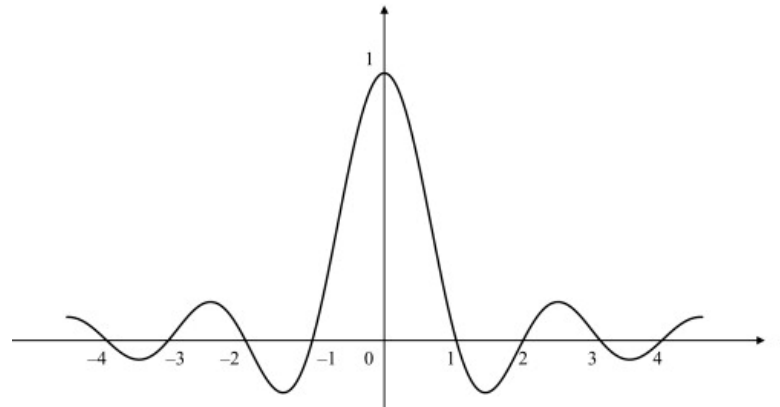
frequency domain



## Aside: Analyzing interpolation methods

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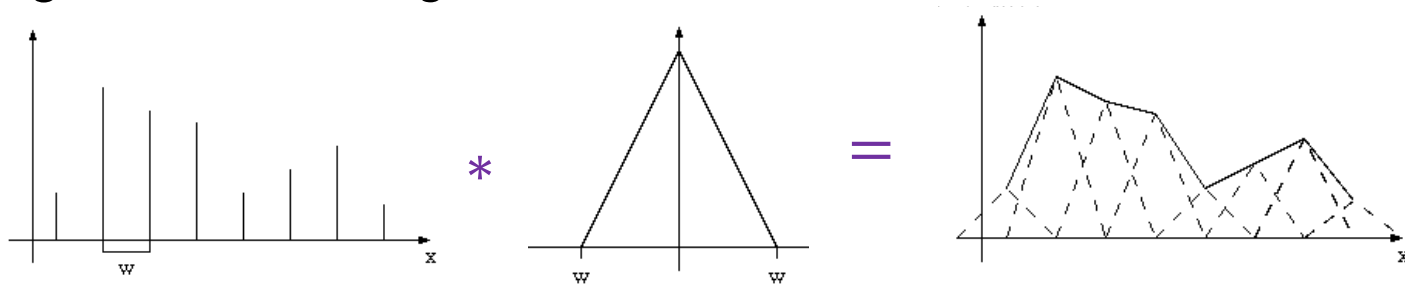
- Perfect reconstruction of the subsampled signal requires convolution with a sinc filter in the spatial domain, which is bad because sinc has infinite support
- Instead, simpler reconstruction (interpolation) methods are typically used



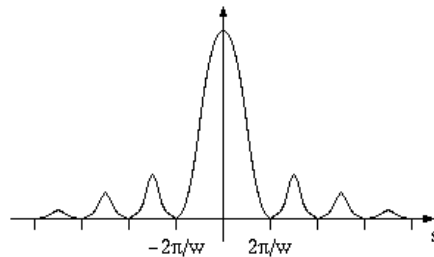
## Aside: Analyzing different interpolation methods

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- Linear reconstruction can be done by convolving the sampled signal with a *triangle filter*:



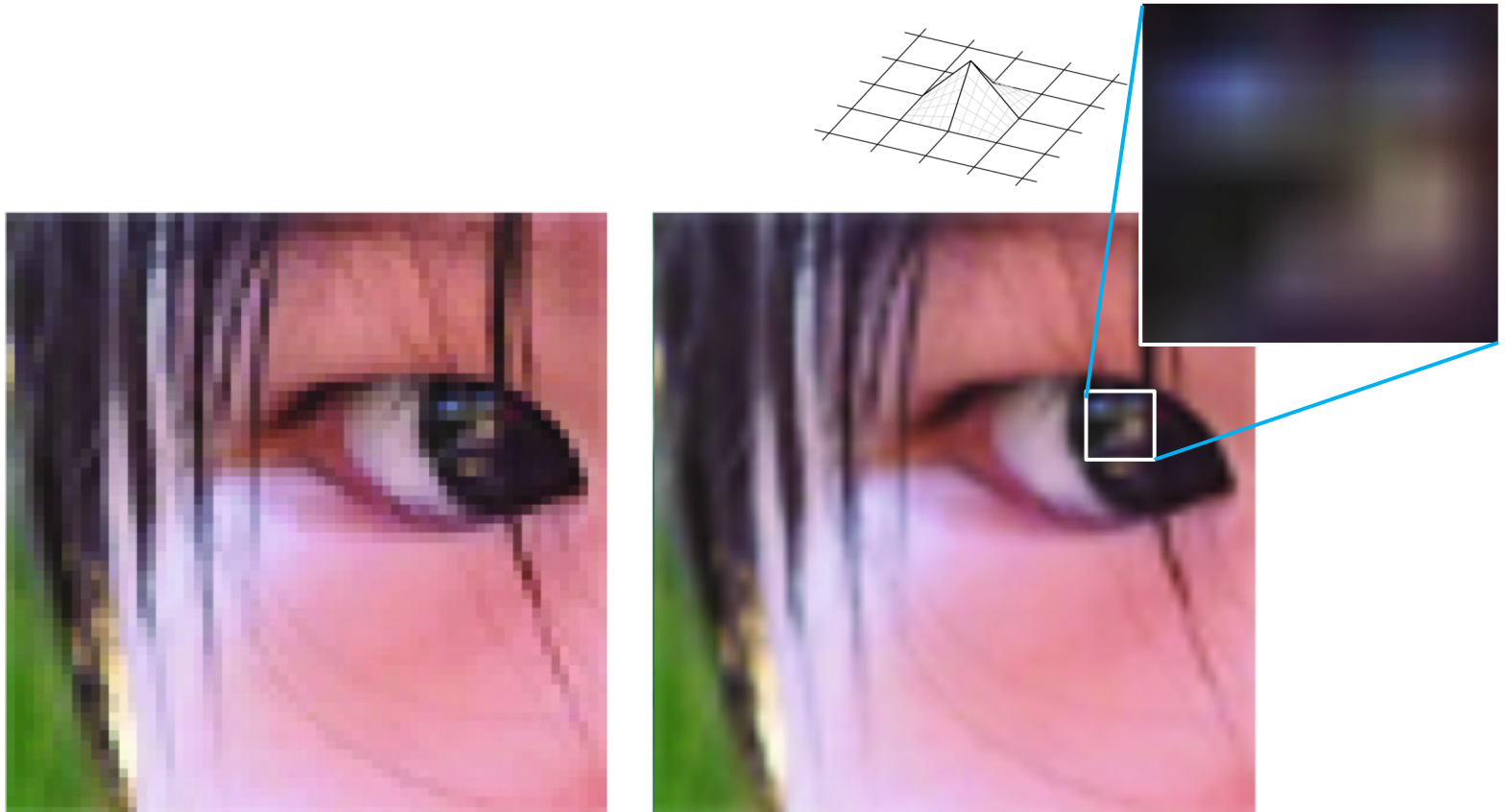
- However, the Fourier transform of the triangle filter is the  $\text{sinc}^2$  function, so multiplying the signal's spectrum by it introduces high-frequency artifacts



[Image source](#)

# Bilinear interpolation closeup

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[Image source](#)



# Why else should you care about Fourier analysis?

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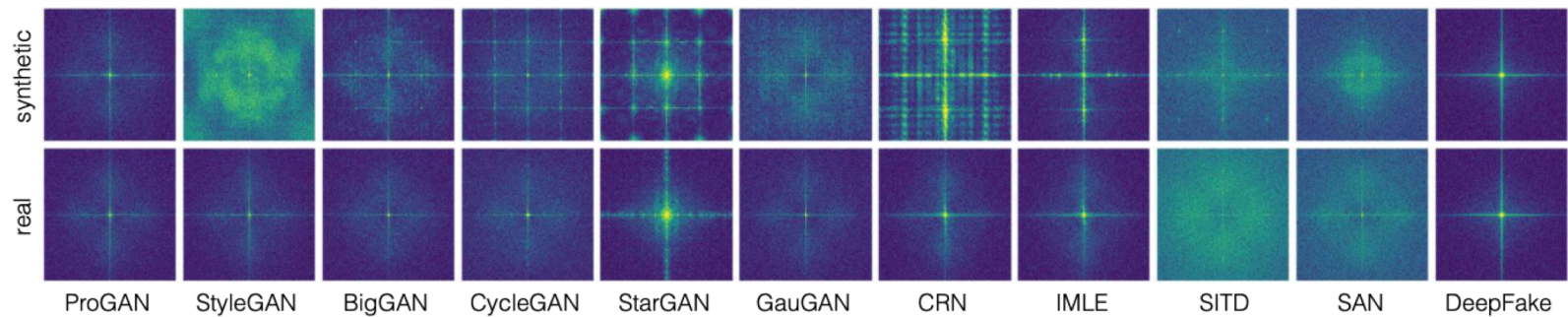


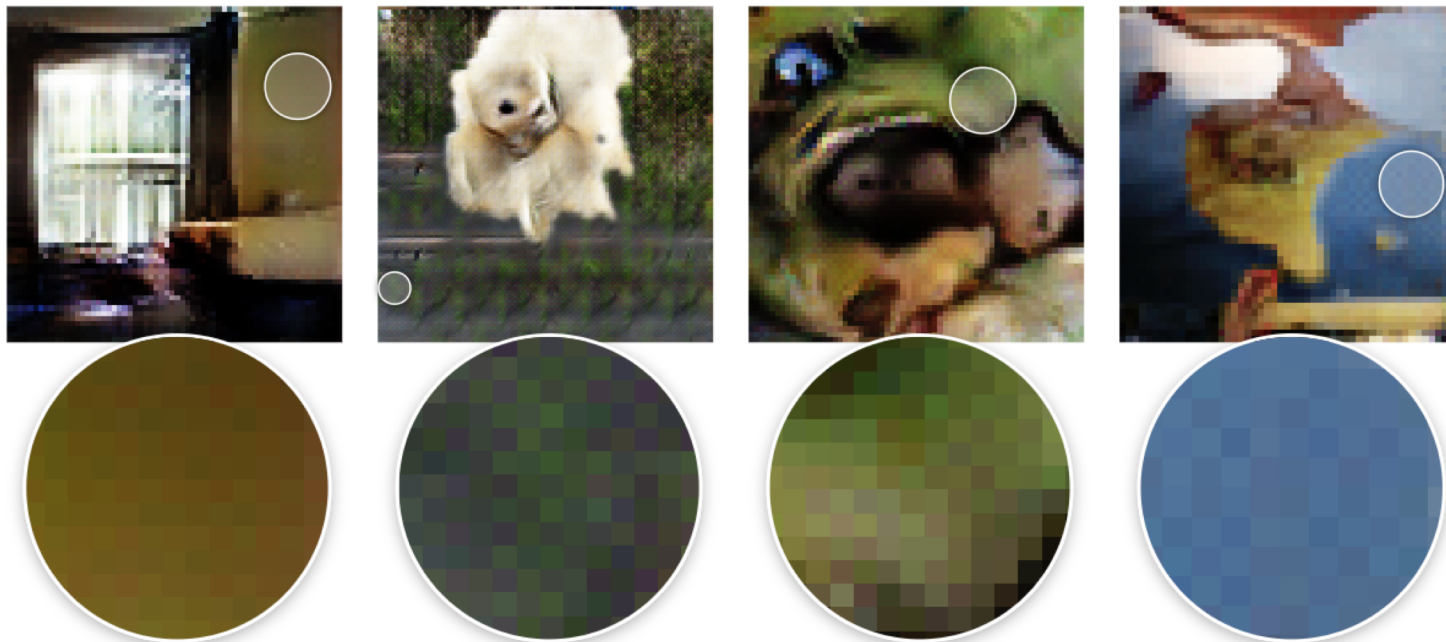
Figure 7: **Frequency analysis on each dataset.** We show the average spectra of each high-pass filtered image, for both the real and fake images, similar to Zhang *et al.* [50]. We observe periodic patterns (dots or lines) in most of the synthetic images, while BigGAN and ProGAN contains relatively few such artifacts.

S.-Y. Wang et al. [CNN-generated images are surprisingly easy to spot... for now.](#) CVPR 2020

# Why else should you care about Fourier analysis?

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Checkerboard and repetition artifacts in GAN-generated images



Radford, et al., 2015 [1]

Salimans et al., 2016 [2]

Donahue, et al., 2016 [3]

Dumoulin, et al., 2016 [4]

<https://distill.pub/2016/deconv-checkerboard/>