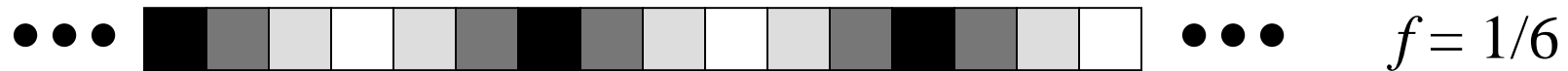


# **Representing Images In Frequency Domain**

# Spatial Frequency Concepts

- General, we use frequency to represent how fast a periodic signal varies over time, expressed in cycles-per-time-step (e.g., Hz for cycles/second).
- Spatial frequency represents how fast the signal (here the pixel values) vary spatially. Consider some 1-D examples:

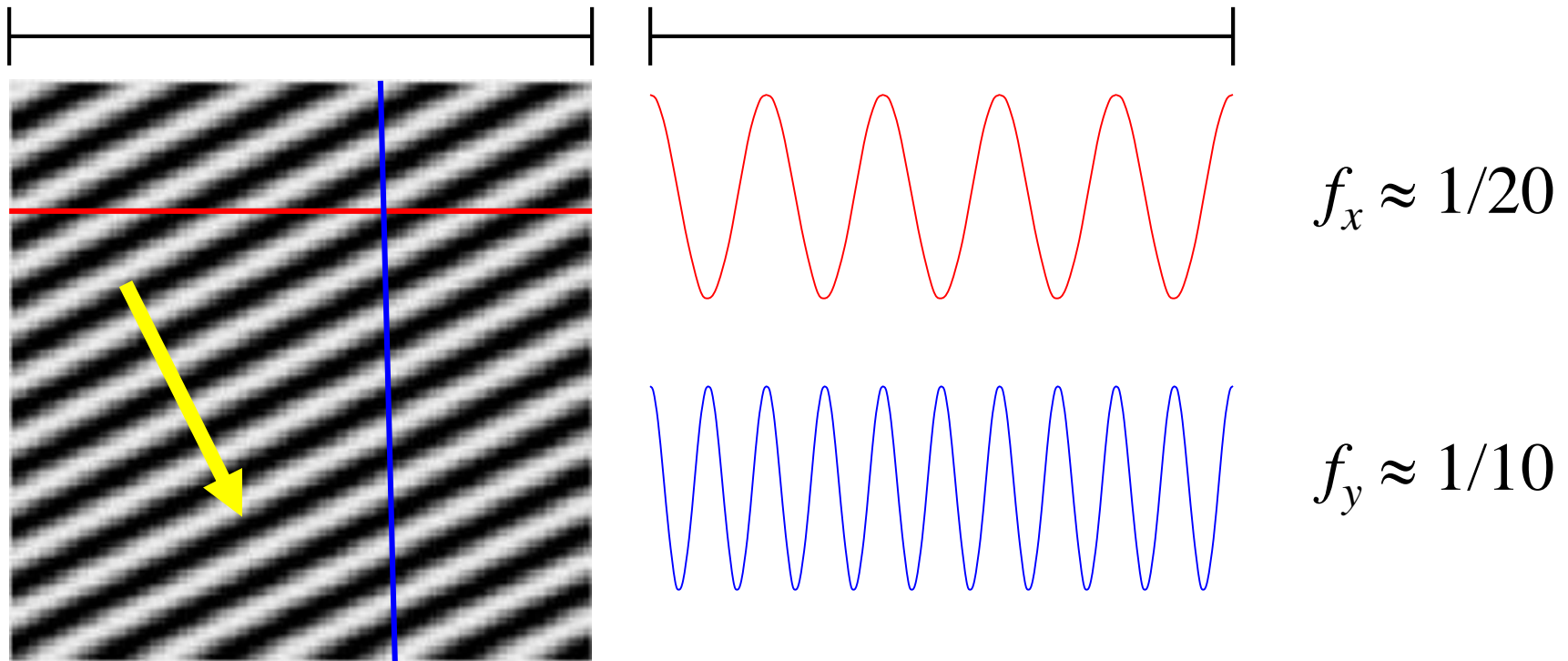


Unit:  
cycles per pixel

# Spatial Frequency Concepts

Spatial frequency for 2-D signals:

- One frequency for each dimension.
- The overall spatial frequency is a vector (wave vector).

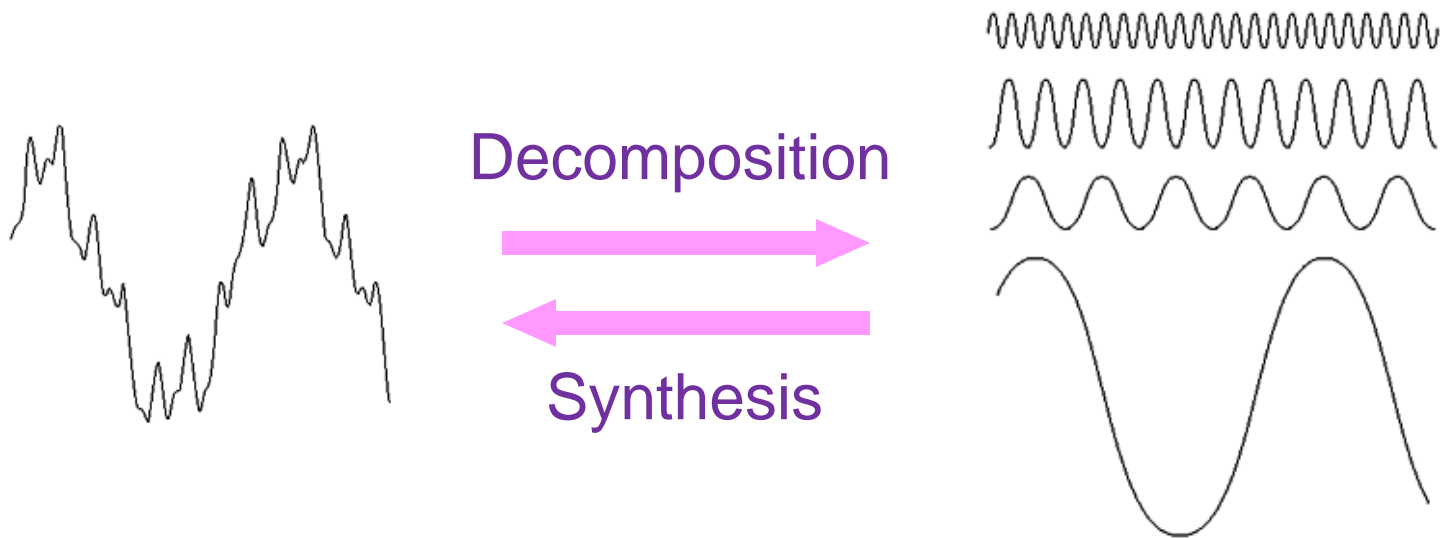


Assuming image size of 100x100

Overall:  $\sqrt{f_x^2 + f_y^2}$

# Spectrum

- In general, the "waveform" of a signal consists of a **mixture** of many sinusoidal waves:



- The information regarding the **frequencies**, **amplitudes**, and **phases** of these sinusoidal waves is collectively called the **spectrum** of the signal.

# Fourier Series

**Fourier Series:** Any periodic function of period  $T$  can be expressed as the combination of a series of sine and/or cosine functions multiplied by appropriate coefficients:



## Synthesis

$$f(t) = \sum_{n=-\infty}^{\infty} \left[ c_n \exp\left(j \frac{2\pi n}{T} t\right) \right] \quad \text{while} \quad f(t) = f(t+T)$$

## Decomposition

To get the (complex) coefficients:  $c_n = \frac{1}{T} \int_{-T/2}^{T/2} f(t) \exp\left(-j \frac{2\pi n}{T} t\right) dt$

This is the projection of the function  $f(t)$  on a set of orthogonal basis functions.

# Continuous Fourier Transform

**Fourier transform:** Any function (periodic or not) with finite total area under the curve can be expressed as the integral of sine and/or cosine functions multiplied by different coefficients:

The Fourier Transform pair:

$$\text{FT: } F(\mu) = \int_{-\infty}^{\infty} f(t) \exp(-j2\pi\mu t) dt$$

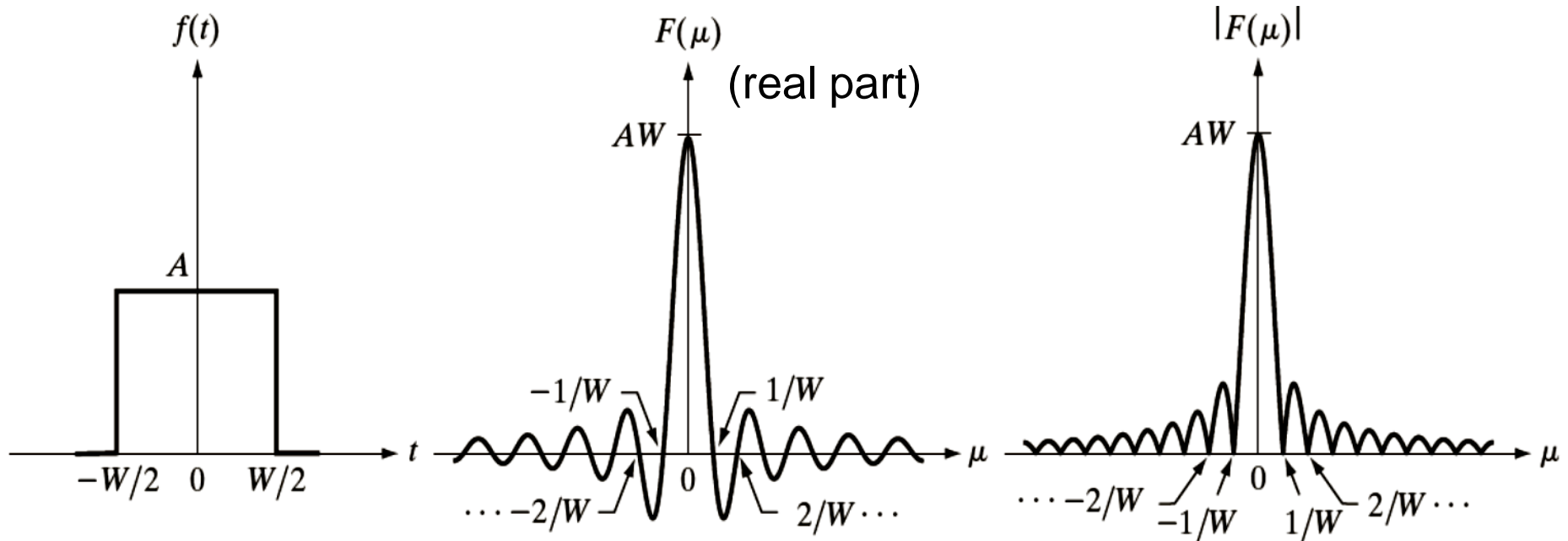
Decomposition

$$\text{IFT: } f(t) = \int_{-\infty}^{\infty} F(\mu) \exp(j2\pi\mu t) d\mu$$

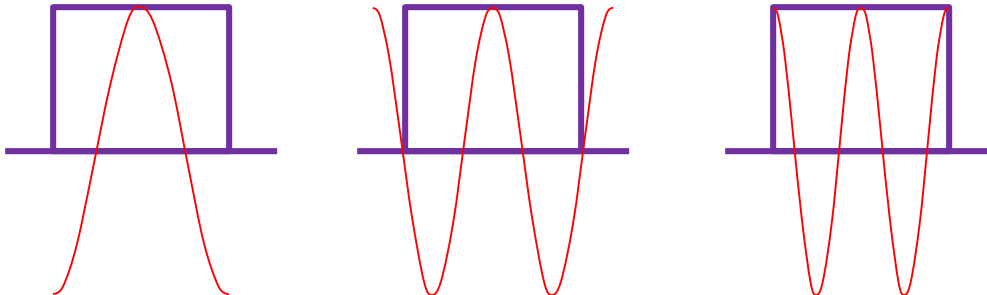
Synthesis

Here  $\mu$  represents frequency (in cycles per unit of space or time).

# Fourier Spectrum



To understand this so-called **sinc** function:



# Fourier Spectrum

- We call the domain of  $\mu$  the frequency domain.
- $F(\mu)$  is complex:  $F(\mu) = R(\mu) + jI(\mu) = |F(\mu)| \exp(j\phi(\mu))$
- **Phase angle**:  $\phi(\mu) = \tan^{-1}[I(\mu) / R(\mu)]$
- The **power spectrum** of  $f(t)$  is  $P(\mu) = |F(\mu)|^2$
- The **spectral density** at a frequency  $\mu$  is just  $|F(\mu)|^2$
- $F(0)$  is the DC term. The other terms are AC terms.
- For real-valued  $f(t)$ :  $F(\mu) = F^*(-\mu)$  and  $|F(\mu)| = |F(-\mu)|$



# Frequency-Domain Image Processing

- The general process:
  - Convert the original image to its frequency domain representation (the spectrum) – Fourier transform.
  - Modify the spectrum.
  - Convert the spectrum back to an image (spatial domain) – Inverse Fourier transform.
- Two main issues for applying frequency-domain processing to images:
  - Spatially discrete (sampled) signals.
  - Finite spatial ranges.

# Delta (Impulse) Function

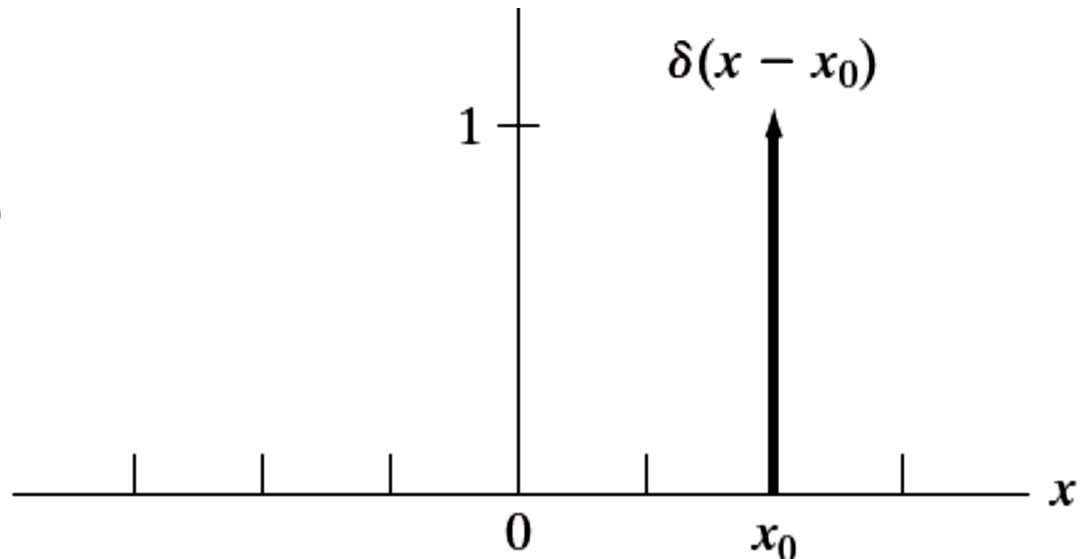
We introduce the concept of **delta functions** here in order to model the sampling operation.

$$\delta(t) = \infty \quad \text{for } t = 0, \quad 0 \quad \text{otherwise}$$

$$\int \delta(t) dt = 1$$

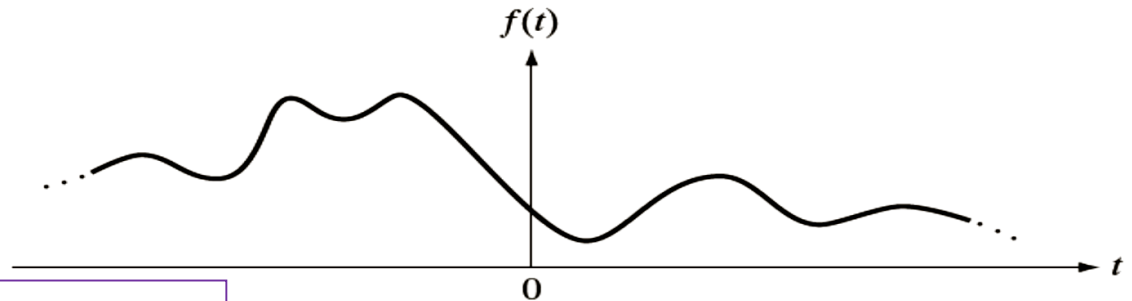
$$\int f(t) \delta(t - t_0) dt = f(t_0)$$

$$\int \exp(j2\pi \mu t) dt = \delta(\mu)$$

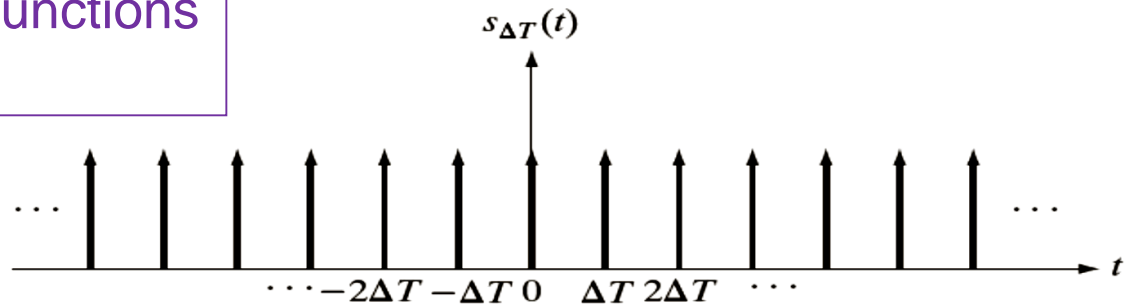


# Sampled Signal

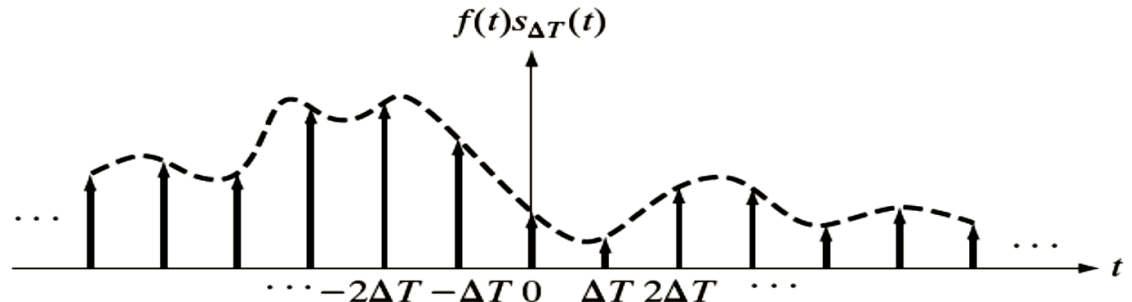
Continuous Signal



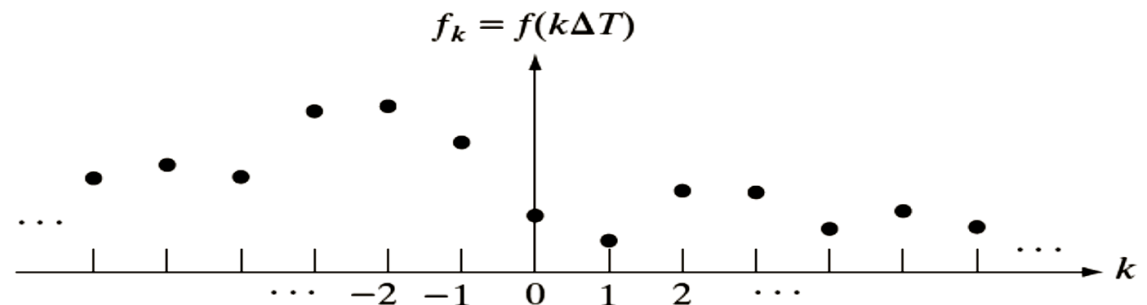
A periodic series of delta functions  
(an impulse train)



Multiplication



Sampled Signal  
(Infinite-height impulses with areas given by the sampled values.)



# FT of Impulse Train

The impulse train for sampling is a periodic function with period of  $\Delta T$ .

$$s_{\Delta T}(t) = \sum_{n=-\infty}^{\infty} \delta(t - n\Delta T)$$

Therefore, we can express it using Fourier series:

$$s_{\Delta T}(t) = \sum_{n=-\infty}^{\infty} \left[ c_n \exp(j \frac{2\pi n}{\Delta T} t) \right] \Rightarrow c_n = \frac{1}{\Delta T}$$
$$\Rightarrow s_{\Delta T}(t) = \frac{1}{\Delta T} \sum_{n=-\infty}^{\infty} \exp(j \frac{2\pi n}{\Delta T} t)$$

By applying FT to this expression of impulse train, we get


$$S(\mu) = \frac{1}{\Delta T} \sum_{n=-\infty}^{\infty} \delta(\mu - \frac{n}{\Delta T})$$

This is an impulse train, with period of  $(\Delta T)^{-1}$ , in the frequency domain.

# Convolution

Continuous Space, 1-D

Convolution: 
$$f(t) * h(t) = \int_{-\infty}^{\infty} f(\tau) h(t - \tau) d\tau$$

 Convolution kernel

**Convolution Theorem:**

$$\phi(t) = f(t) * h(t) \quad \Leftrightarrow \quad \Phi(\mu) = F(\mu) H(\mu)$$

$$\phi(t) = f(t) h(t) \quad \Leftrightarrow \quad \Phi(\mu) = F(\mu) * H(\mu)$$

Convolution theorem allows us to understand the effect of temporal/spatial domain processing in the frequency domain, and vice versa.

We introduce convolution theorem here in order to derive the FT of sampled signals.

# FT of Sampled Signal

Sampled Signal

$$\tilde{f}(t) = f(t)s_{\Delta T}(t)$$



Convolution theorem

FT of Sampled Signal

$$\tilde{F}(\mu) = F(\mu) * S(\mu) = \int_{-\infty}^{\infty} F(\tau)S(\mu - \tau) d\tau$$

$$= \frac{1}{\Delta T} \int_{-\infty}^{\infty} F(\tau) \sum_{n=-\infty}^{\infty} \delta\left(\mu - \tau - \frac{n}{\Delta T}\right) d\tau$$

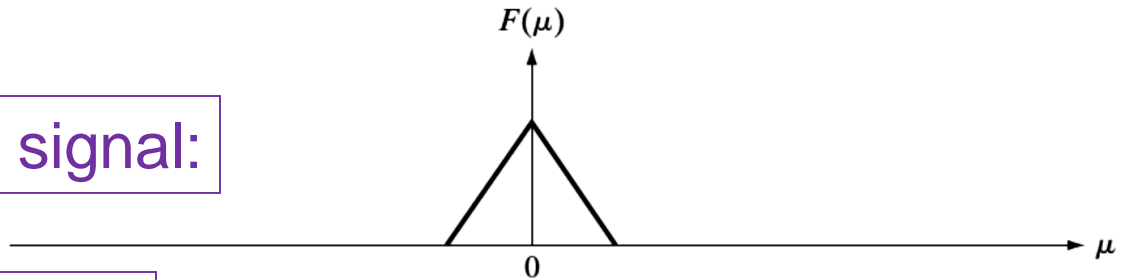
$$= \frac{1}{\Delta T} \sum_{n=-\infty}^{\infty} \int_{-\infty}^{\infty} F(\tau) \delta\left(\mu - \tau - \frac{n}{\Delta T}\right) d\tau$$

$$= \frac{1}{\Delta T} \sum_{n=-\infty}^{\infty} F\left(\mu - \frac{n}{\Delta T}\right)$$

We get multiple copies of  $F(\mu)$ , FT of the continuous signal, evenly spaced by  $(\Delta T)^{-1}$ .

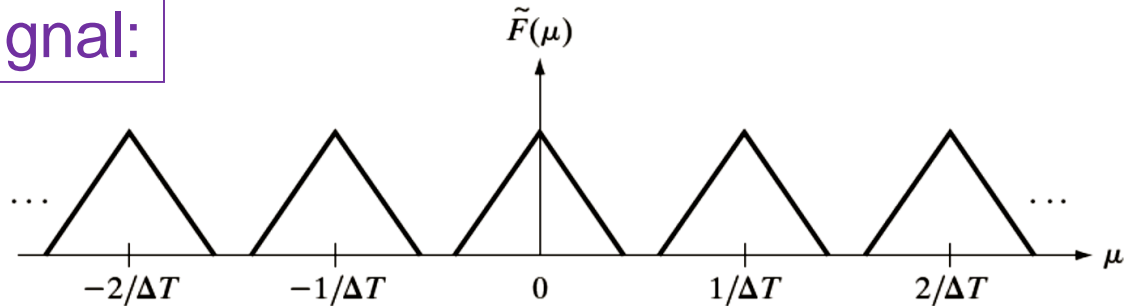
# FT of Sampled Signal: Examples

Spectrum of continuous signal:

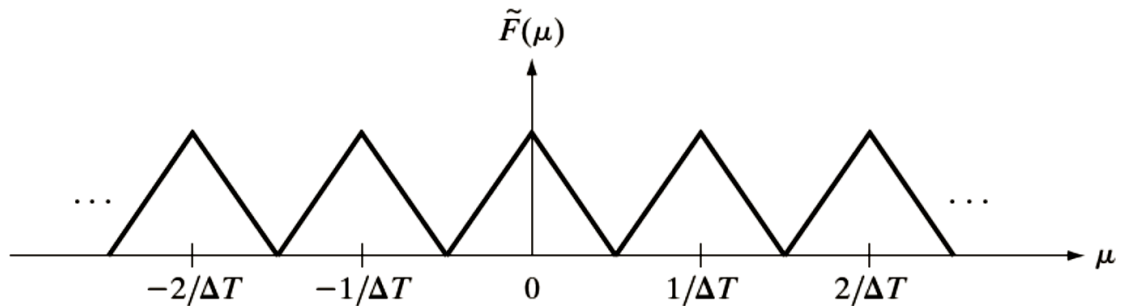


Spectrum of sampled signal:

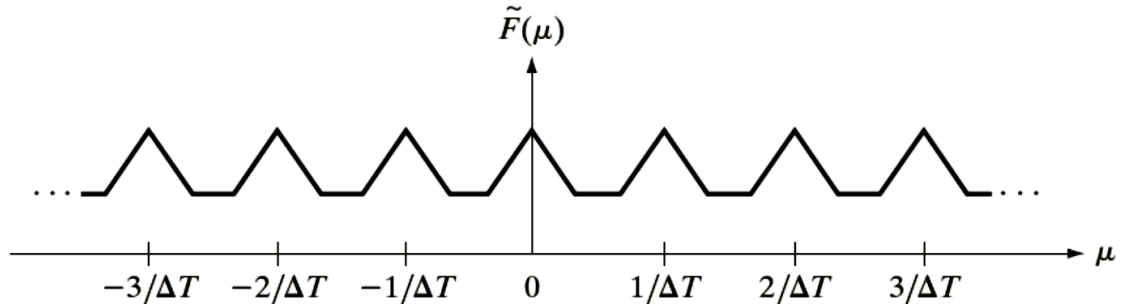
Over-sampling:



Critically-sampling:

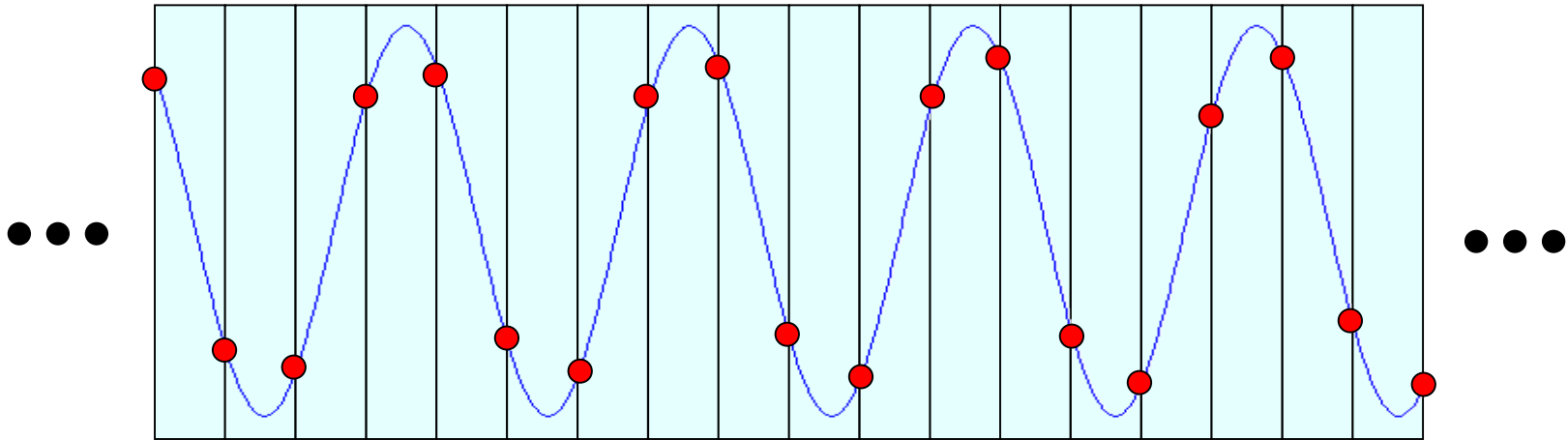


Under-sampling:

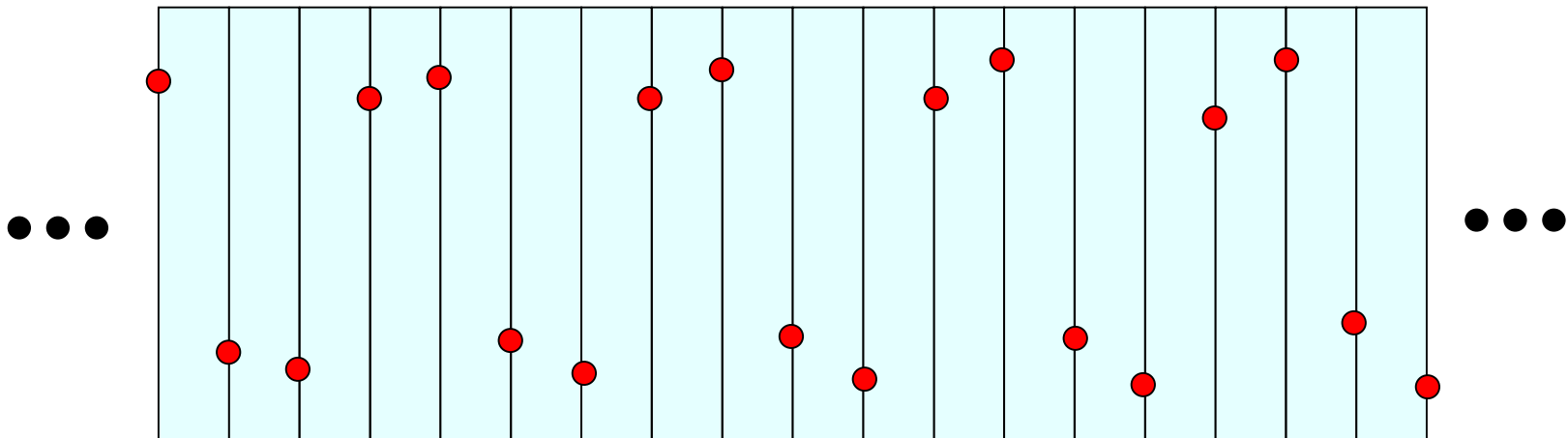


# Illustrations of Sampling

Sampling a sinusoidal function:

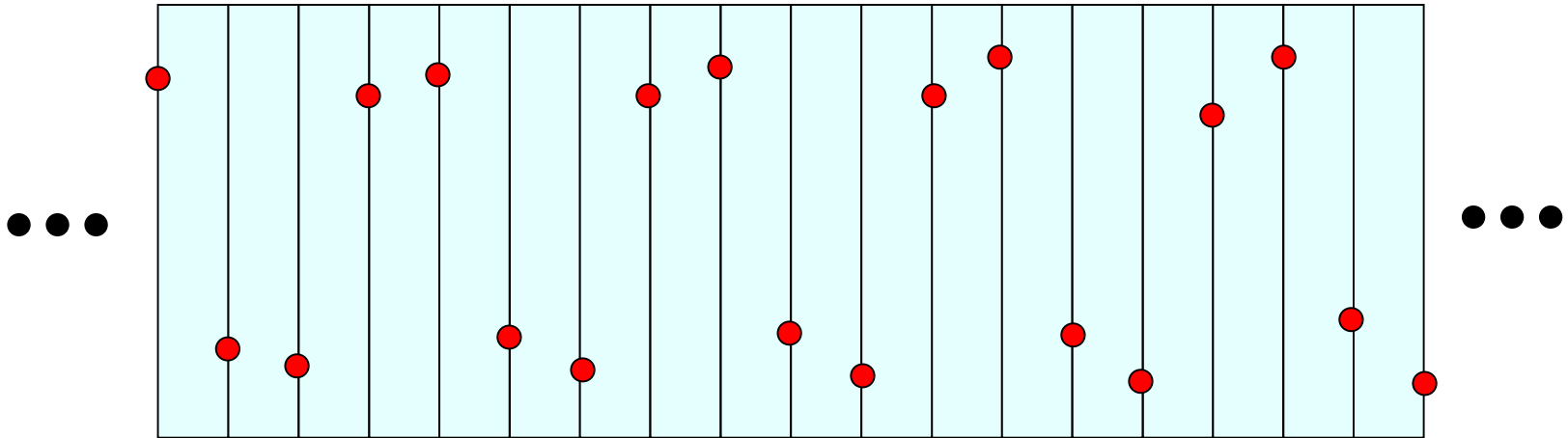


Assume that sampling is done on a 1-D regular grid. We get the sampled values:

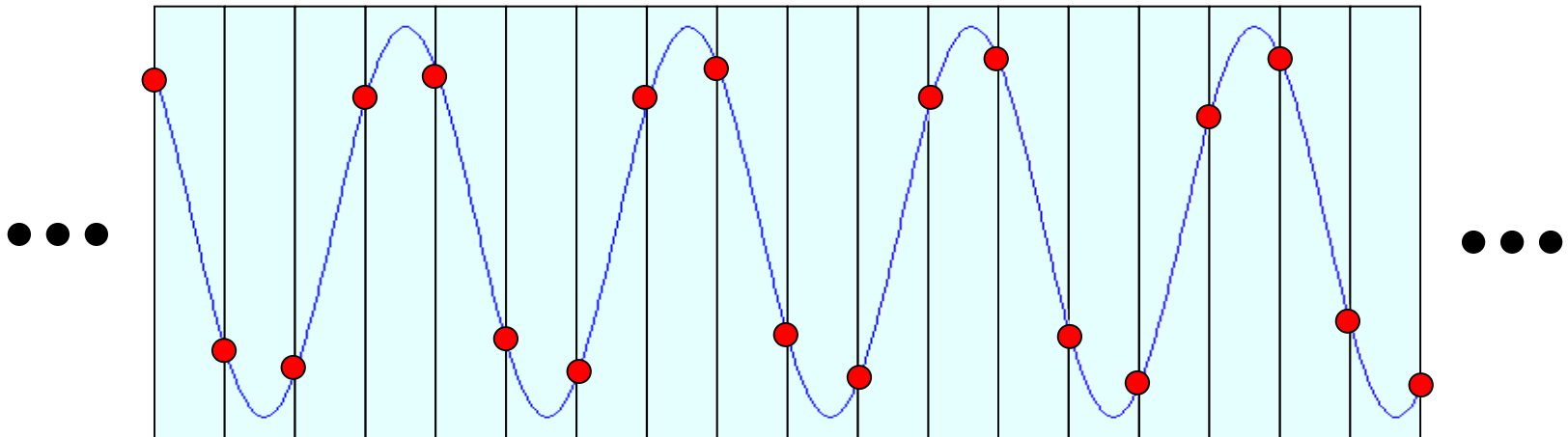




# Illustrations of Sampling

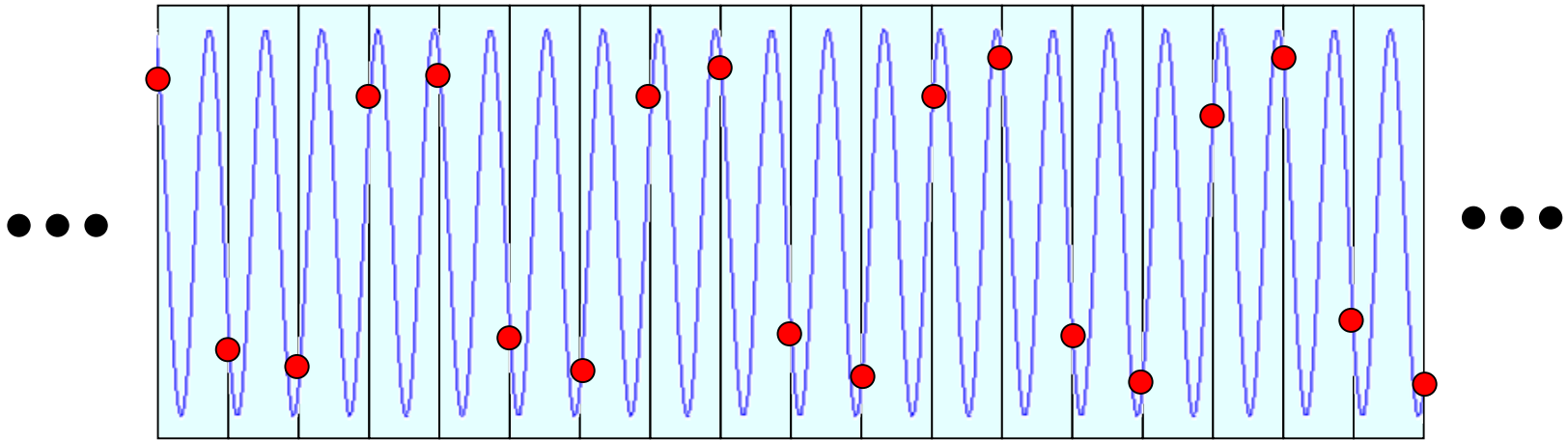


Now if we try to "fit" a sinusoidal function to these sampled points, we can expect to get this, which is the original signal:

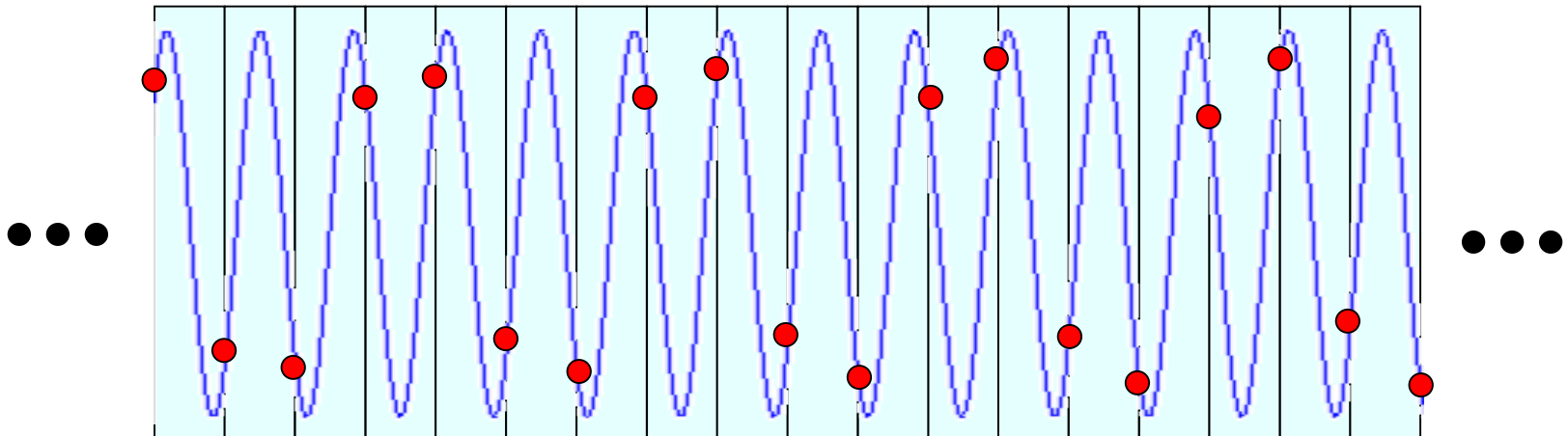


# Illustrations of Sampling

However, we can also fit the sampled data to another sinusoidal function of much higher frequency:

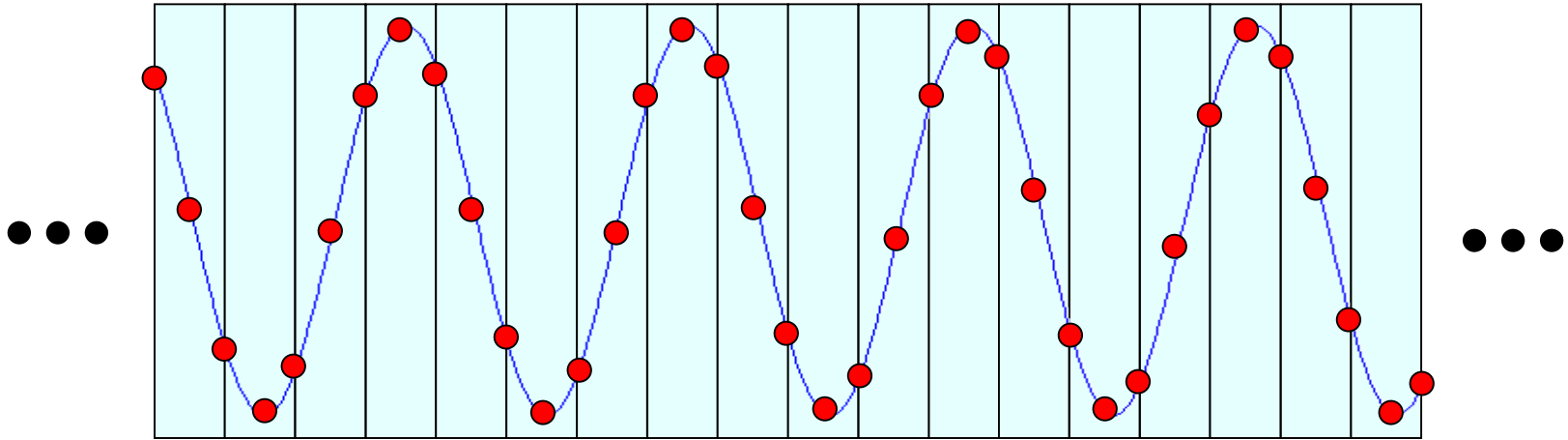


This is yet another solution:

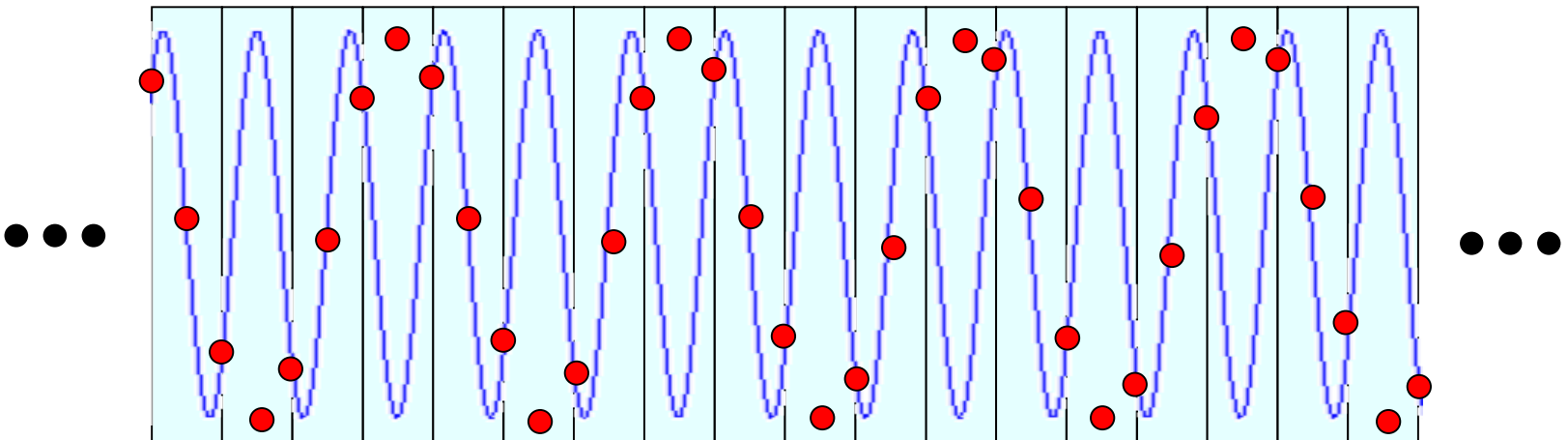


# Illustrations of Sampling

Now, if we double the sampling rate:



We won't get this solution ...



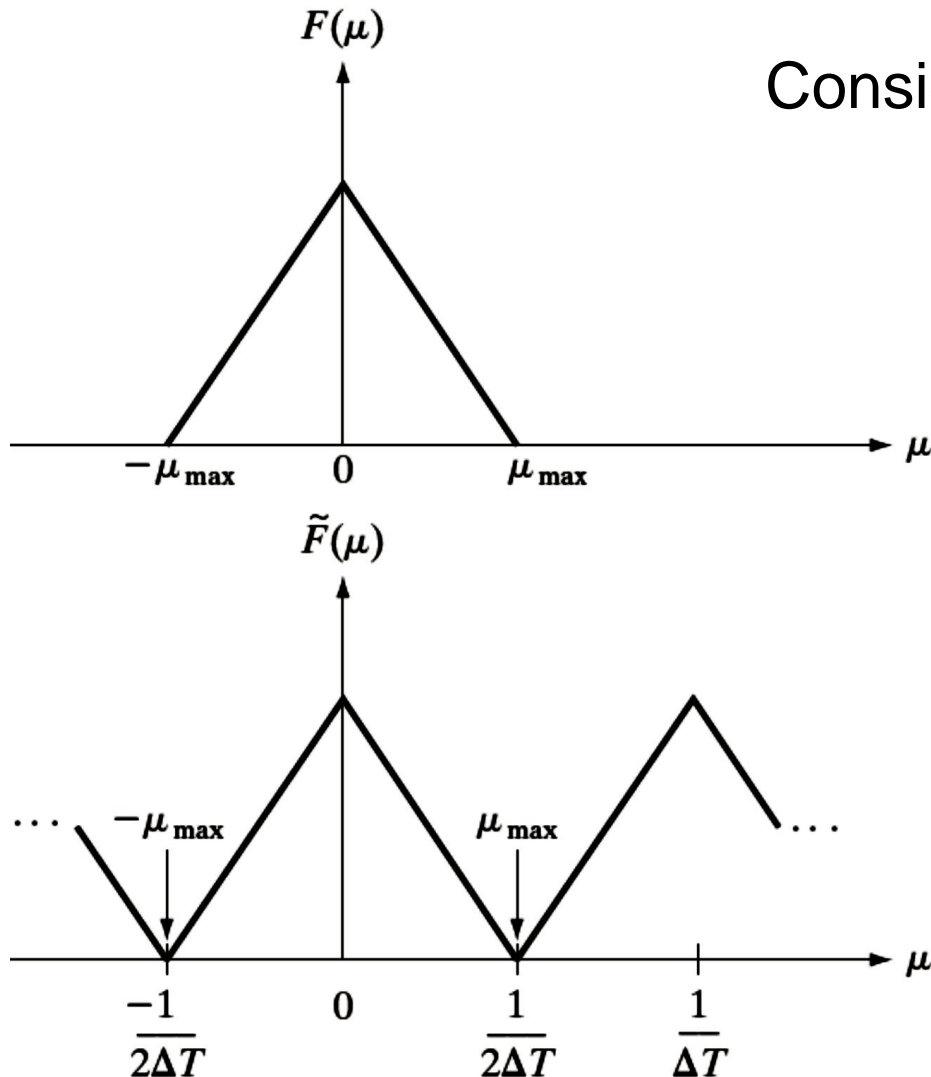
# Sampling Theorem

Consider a **band-limited** function  $f(x)$ :

To avoid overlapping between "copies" of the original  $F(\mu)$ , it is necessary that

$$\frac{1}{\Delta T} \geq 2\mu_{\max}$$

In other words, we need at least two samples per period in the original signal.

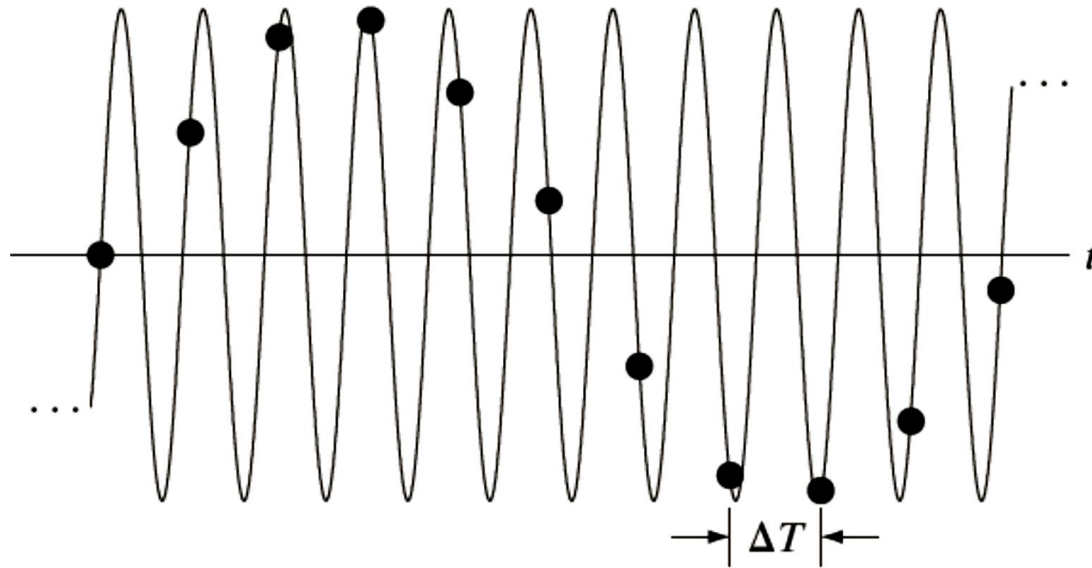


# Aliasing

Simply said, **aliasing** is what happens when the condition

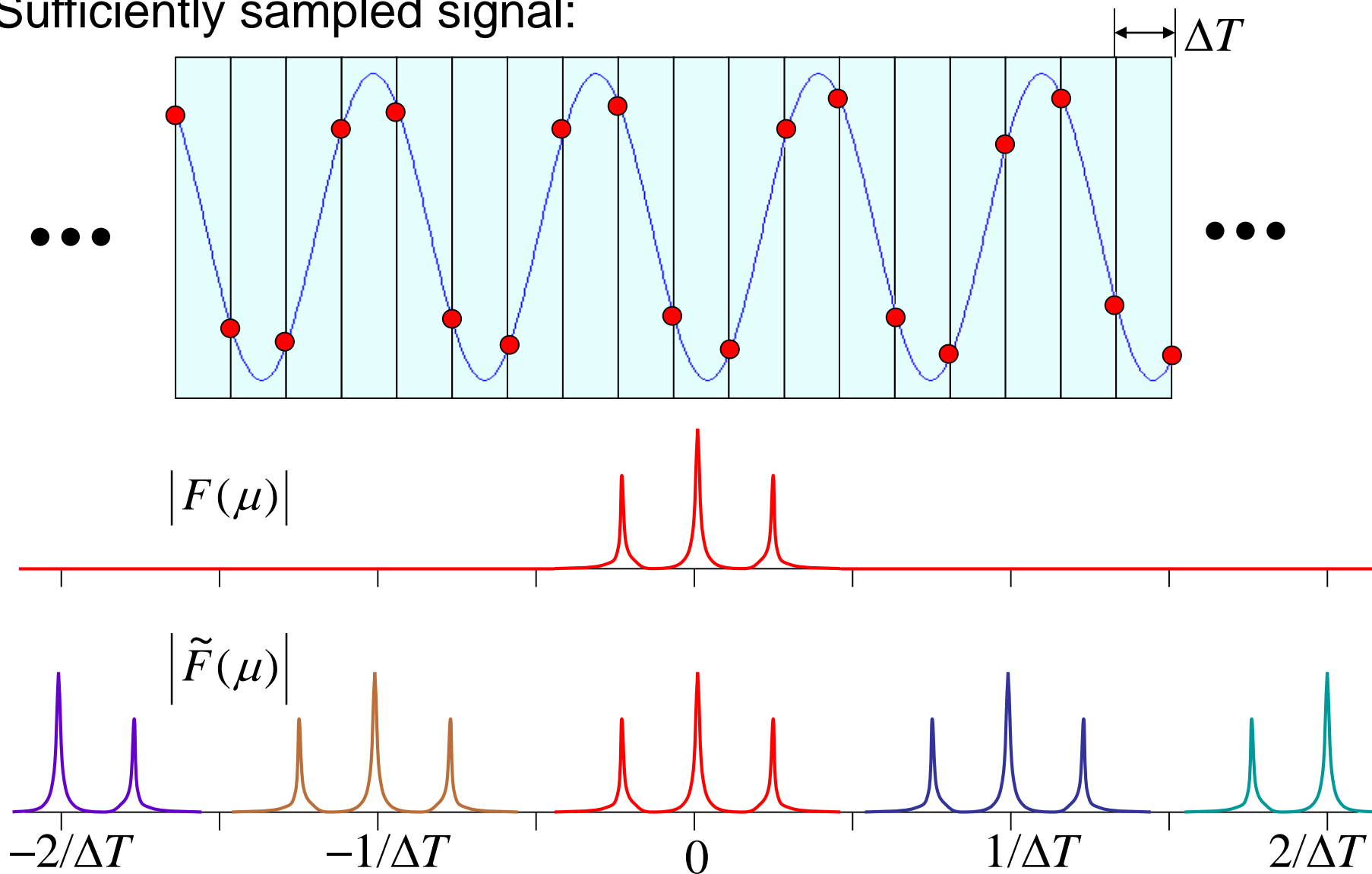
$$\frac{1}{\Delta T} \geq 2\mu_{\max}$$

is not satisfied. We can not faithfully reconstruct the original signal from the sampled data in this case.



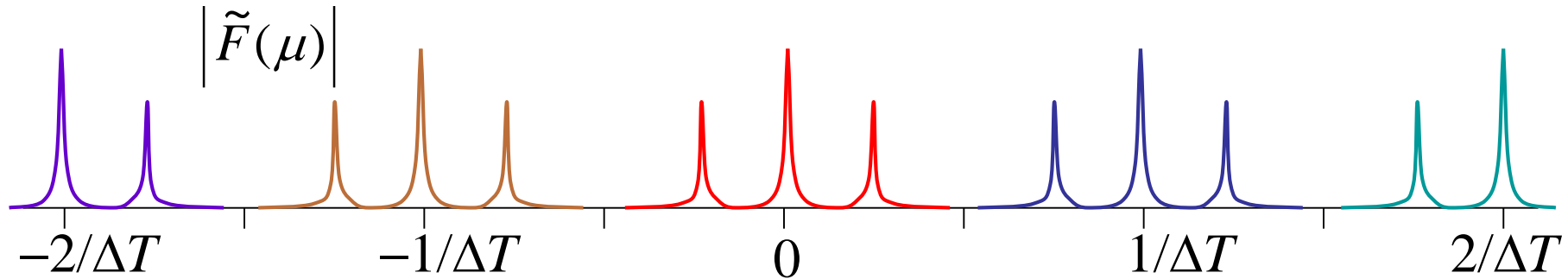
# Aliasing – More Illustrations

Sufficiently sampled signal:



# Aliasing – More Illustrations

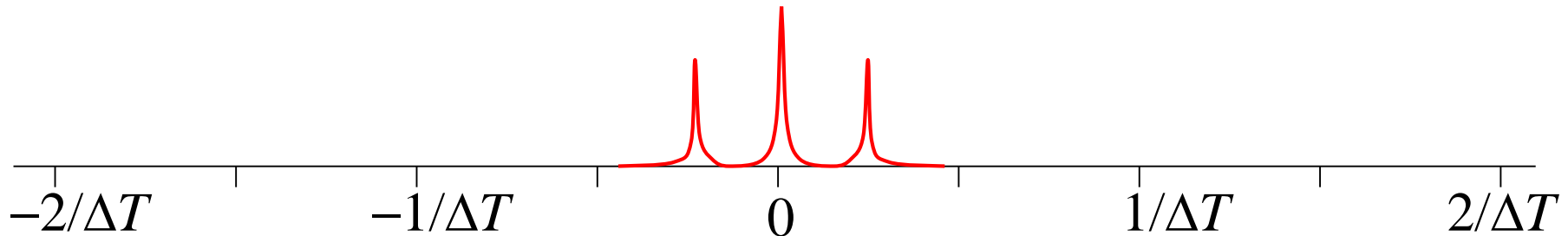
Reconstruction of the sufficiently sampled signal:



Limit the spectrum to include only one copy of  $F(\mu)$  by multiplying with this:

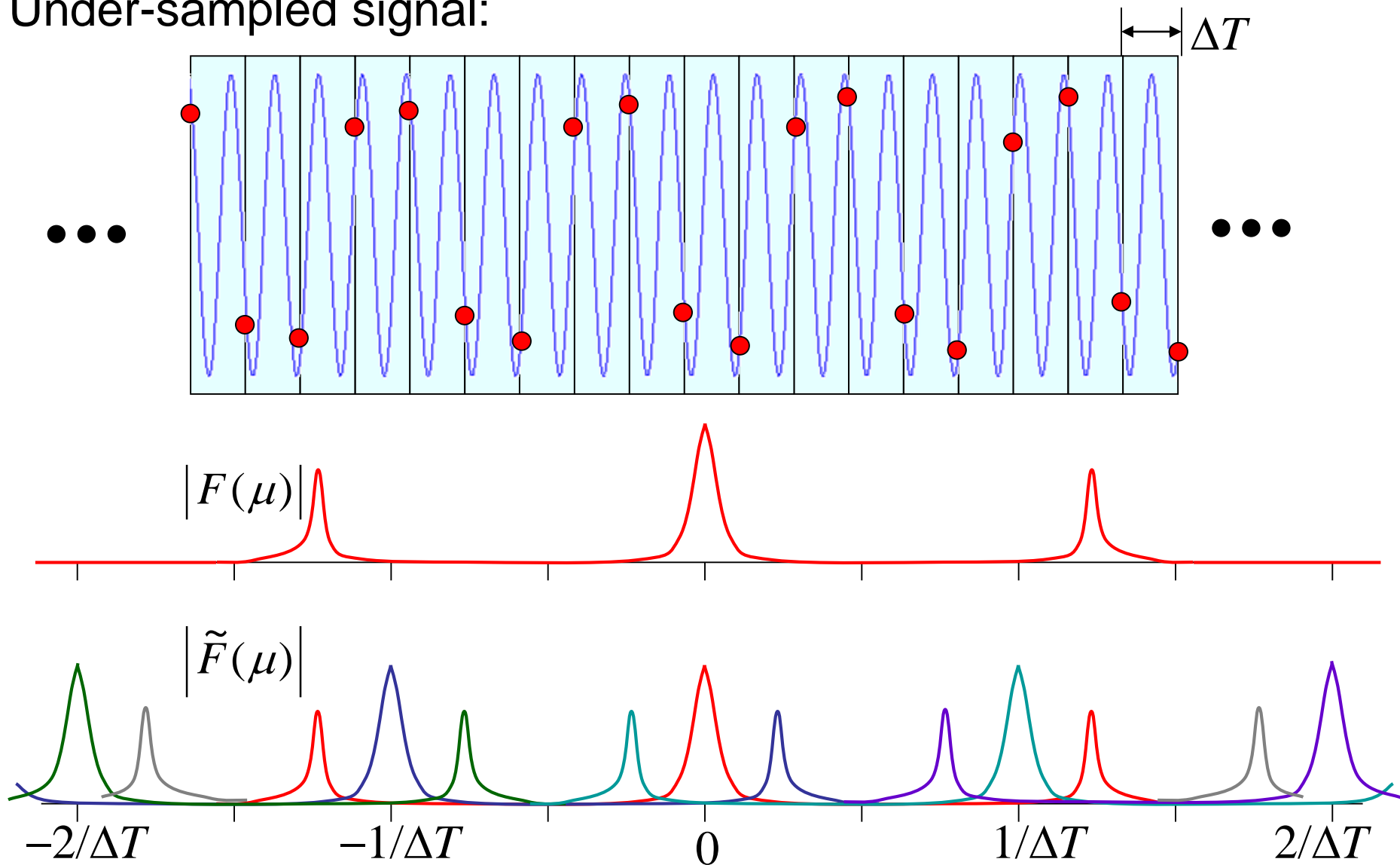


This is what is used for reconstruction (correct signal frequency):



# Aliasing – More Illustrations

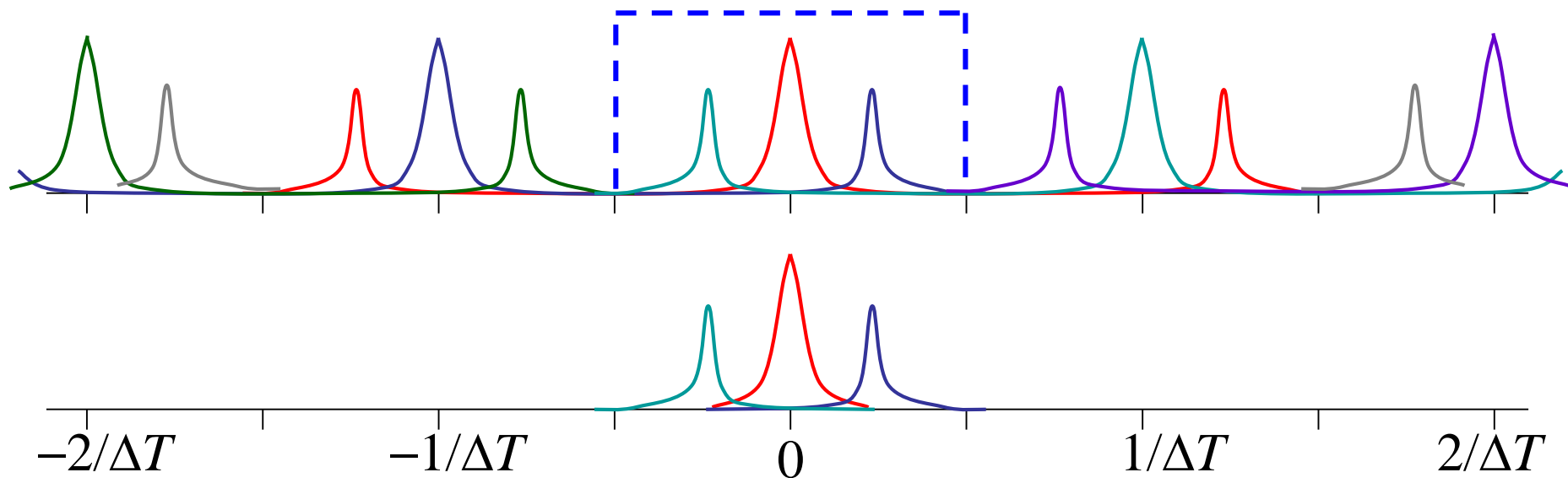
Under-sampled signal:



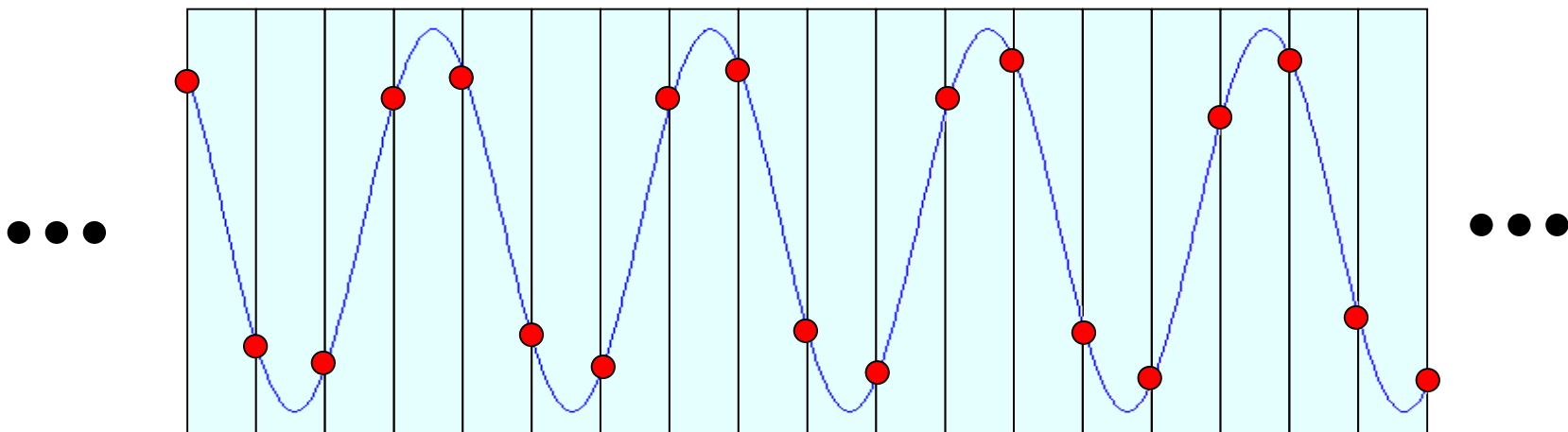


# Aliasing – More Illustrations

Reconstruction of the under-sampled signal:



The reconstructed signal:

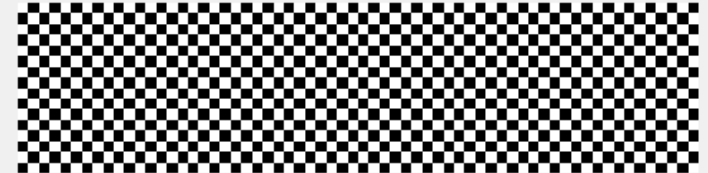


# Aliasing: Examples

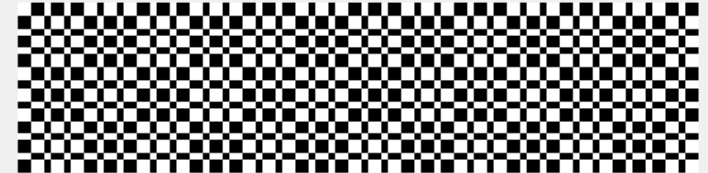
Sinusoidal Wave Signal

Square Wave Signal

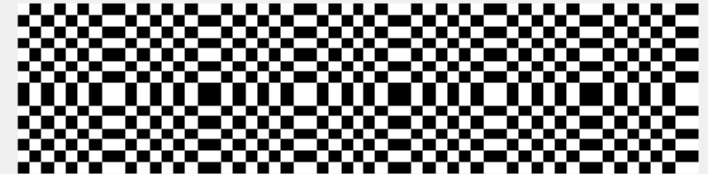
Original  
( $T=16$ )



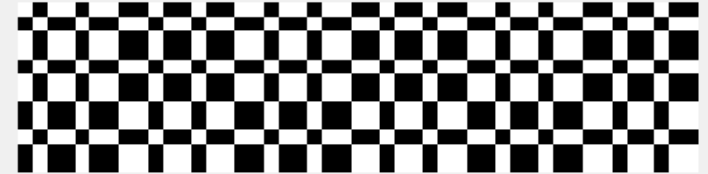
Sampled  
( $\Delta T=5$ )



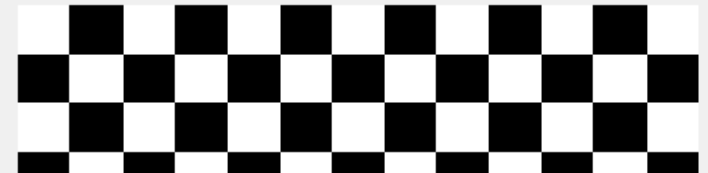
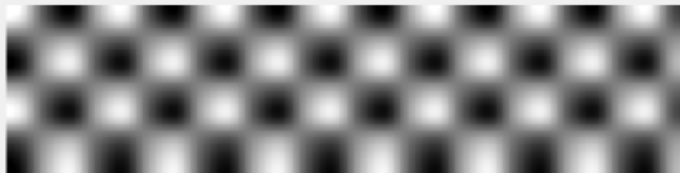
Sampled  
( $\Delta T=9$ )



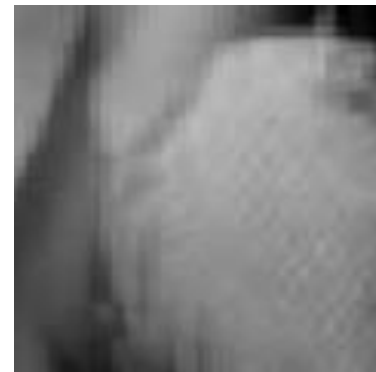
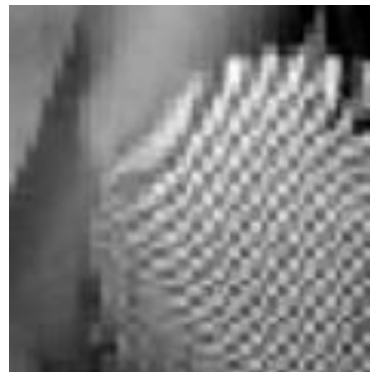
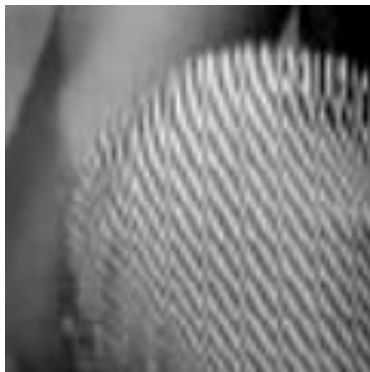
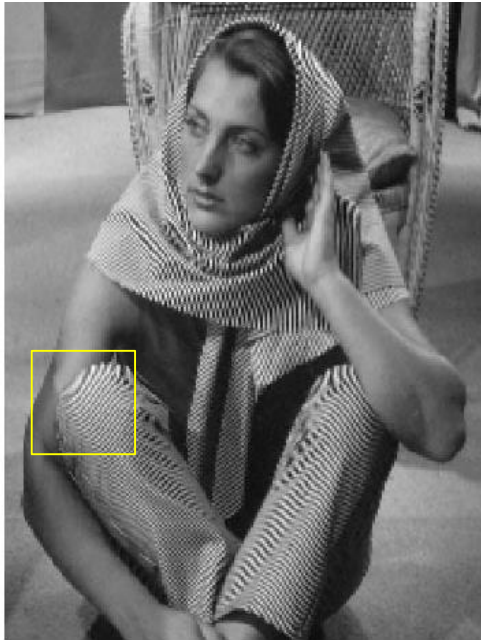
Sampled  
( $\Delta T=11$ )



Sampled  
( $\Delta T=20$ )



# Anti-aliasing by Smoothing



Original

50% size  
reduction

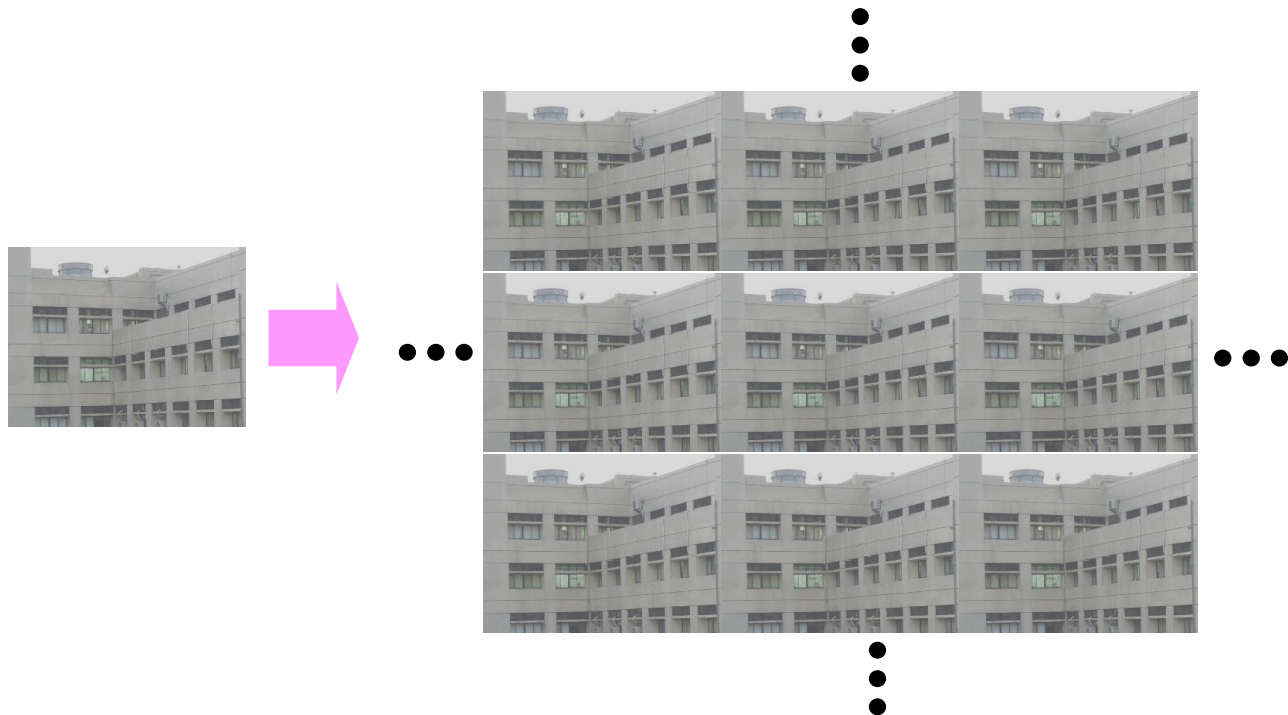
50% size reduction  
after smoothing

# Discrete Fourier Transform (DFT)

Since now we have finite spatial ranges, we can think of discrete Fourier transform as the Fourier series of a periodic function consisting of repeating copies of the original finite-range sampled signal (i.e., the image).



**2-D:**



# Summary of 1-D DFT and IDFT

We consider the FT of  $f(x)$  defined at a finite (total  $M$  samples), discrete, and evenly-spaced set of  $x$  in its domain

$$\text{DFT: } F(u) = \sum_{x=0}^{M-1} f(x) \exp(-j2\pi ux / M), \quad u = 0, 1, \dots, M-1$$

$$\text{IDFT: } f(x) = \frac{1}{M} \sum_{u=0}^{M-1} F(u) \exp(j2\pi ux / M), \quad x = 0, 1, \dots, M-1$$

Note that  $f(x)$  and  $F(u)$  have the same number of elements.

Since both  $f(x)$  and  $F(u)$  are periodic (with period  $M$ ), we only need  $M$  points in each domain for computing the transforms.

# Deriving DFT from Fourier Series

Slightly modified equations of Fourier series (to make the results consistent with the textbook):

$$\tilde{f}(t) = \frac{1}{M} \sum_{u=-\infty}^{\infty} \left[ F(u) \exp\left(j \frac{2\pi ut}{M}\right) \right]$$

Here  $t$  is a continuous variable. Both  $x$  and  $u$  are discrete (integral) variables.

$$F(u) = \int_0^{M-} \tilde{f}(t) \exp\left(-j \frac{2\pi ut}{M}\right) dt$$

$$= \int_0^{M-} f(t) \sum_{x=-\infty}^{\infty} \delta(t - x) \exp\left(-j \frac{2\pi ut}{M}\right) dt$$

$$= \sum_{x=0}^{M-1} f(x) \exp\left(-j \frac{2\pi ux}{M}\right)$$



This is periodic:  
 $F(u + M) = F(u)$

# Proof of Inverse DFT

$$\begin{aligned} & \frac{1}{M} \sum_{u=0}^{M-1} F(u) \exp\left(j \frac{2\pi ux}{M}\right) \\ &= \frac{1}{M} \sum_{u=0}^{M-1} \sum_{x'=0}^{M-1} f(x') \exp\left(-j \frac{2\pi ux'}{M}\right) \exp\left(j \frac{2\pi ux}{M}\right) \\ &= \frac{1}{M} \sum_{x'=0}^{M-1} f(x') \sum_{u=0}^{M-1} \exp\left(-j \frac{2\pi ux'}{M}\right) \exp\left(j \frac{2\pi ux}{M}\right) = f(x) \end{aligned}$$

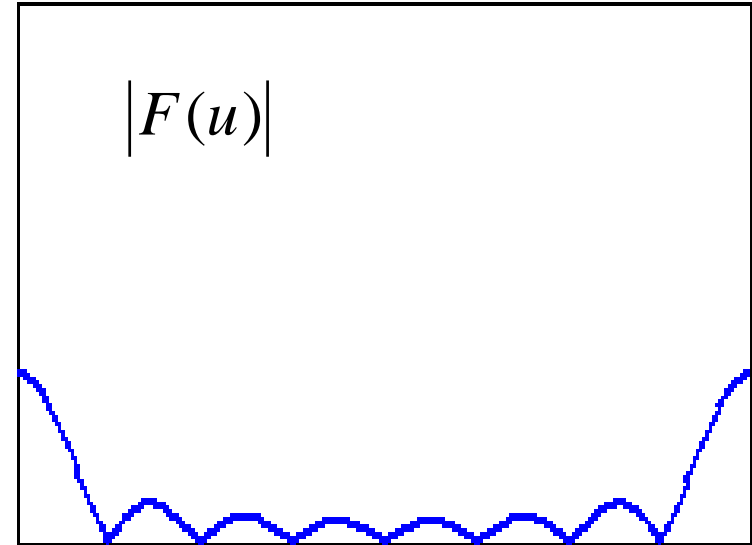
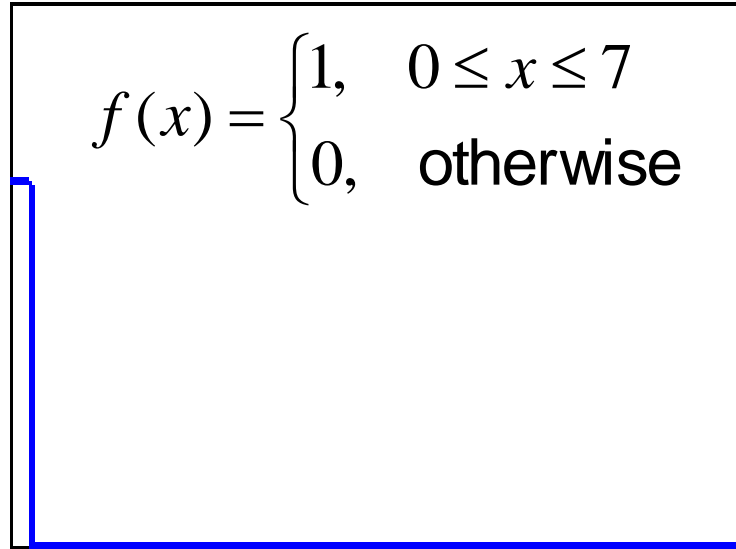
Note:

$$\sum_{u=0}^{M-1} \exp\left(-j \frac{2\pi ux'}{M}\right) \exp\left(j \frac{2\pi ux}{M}\right) = \begin{cases} M & \text{for } x = x' \\ 0 & \text{otherwise} \end{cases}$$

$$x, x' = 0, 1, \dots, M-1$$

# Spectrum - Example

$M=256$



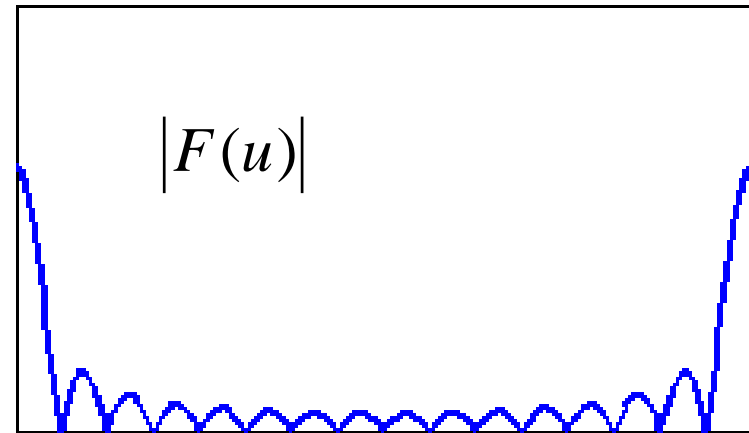
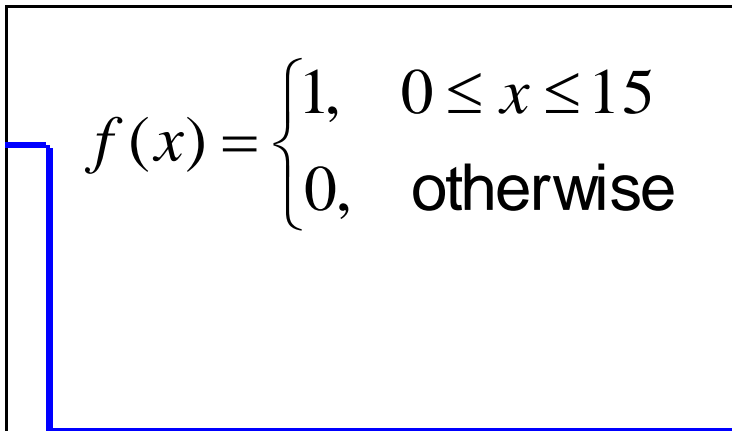
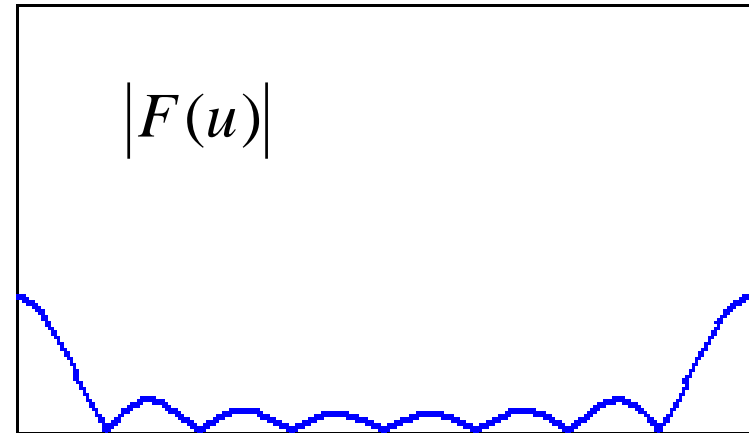
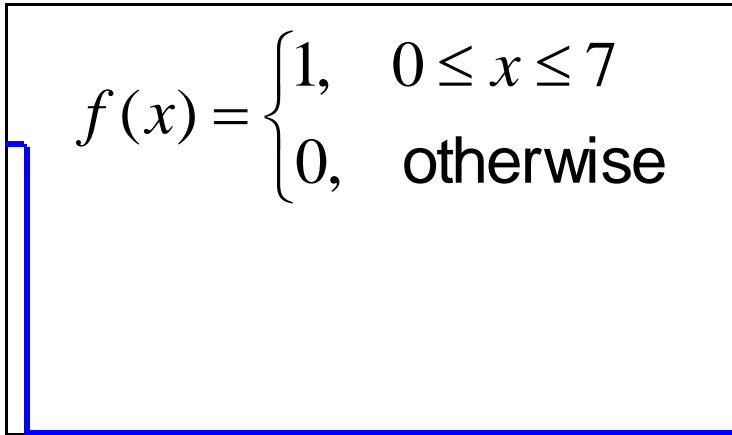
Interesting points:

- What happens at  $|F(0)|$  ?
- What happens at  $|F(M-1)|$  ?
- What happens if we change  $M$  but leaves  $f(x)$  unchanged?
- What happens if we just shift the non-zero part of  $f(x)$  all by a constant  $\Delta x$ ?



# Space-Frequency Reciprocity

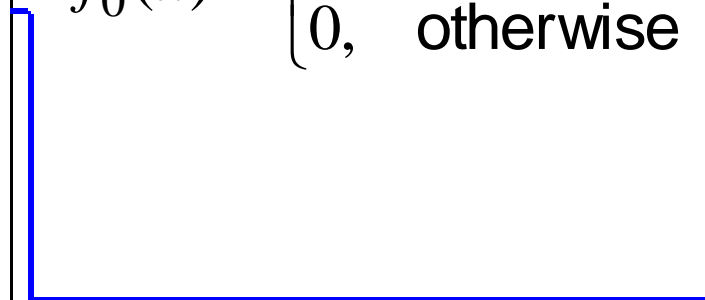
$M=256$




Note in  $|F(u)|$ : (1) peak height, (2) peak width, (3) # zeros

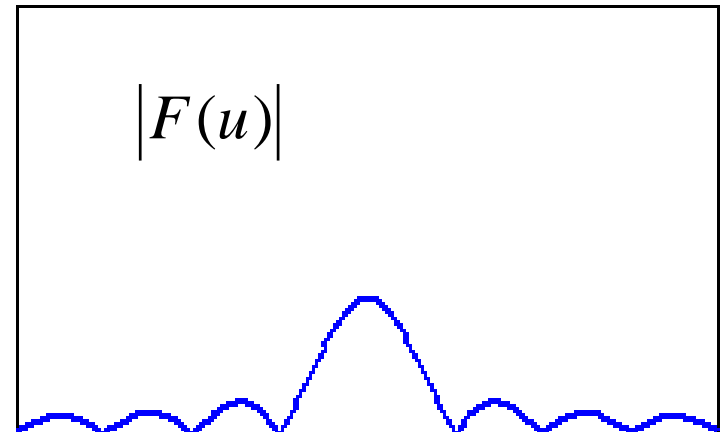
# Centering the Spectrum

When visualizing the spectrum, it is a common practice to multiply  $f(x)$  by  $(-1)^x$  first before applying DFT, so that the DC term of  $|F(u)|$  appears at the center ( $M/2$ ) of the resulting spectrum.

$$f_0(x) = \begin{cases} 1, & 0 \leq x \leq 7 \\ 0, & \text{otherwise} \end{cases}$$




$$f(x) = (-1)^x f_0(x)$$




# 2-D DFT and IDFT

The discrete 2-D FT/IFT (sample size  $M \times N$ ):

$$\text{DFT:} \quad F(u, v) = \sum_{x=0}^{M-1} \sum_{y=0}^{N-1} f(x, y) \exp[-j2\pi(ux/M + vy/N)]$$

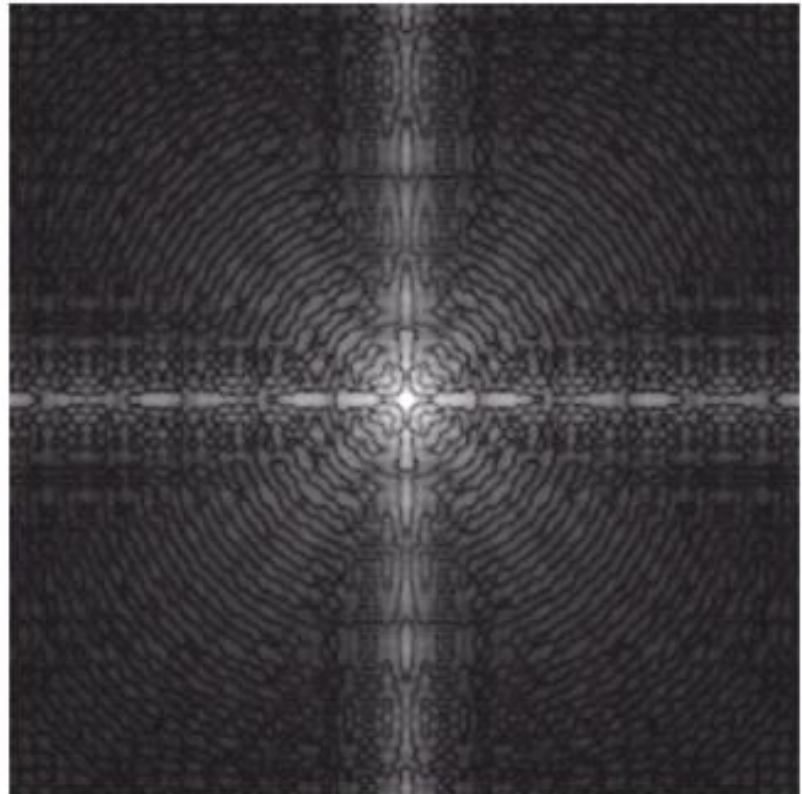
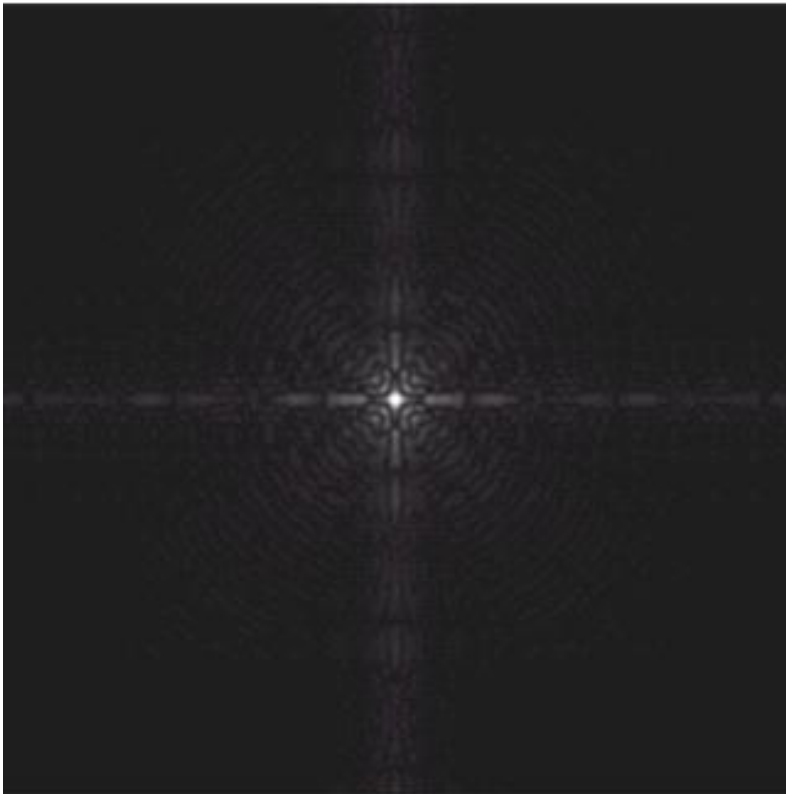
$$\text{IDFT:} \quad f(x, y) = \frac{1}{MN} \sum_{u=0}^{M-1} \sum_{v=0}^{N-1} F(u, v) \exp[j2\pi(ux/M + vy/N)]$$

To center the spectrum: multiply  $f(x, y)$  by  $(-1)^{x+y}$ .

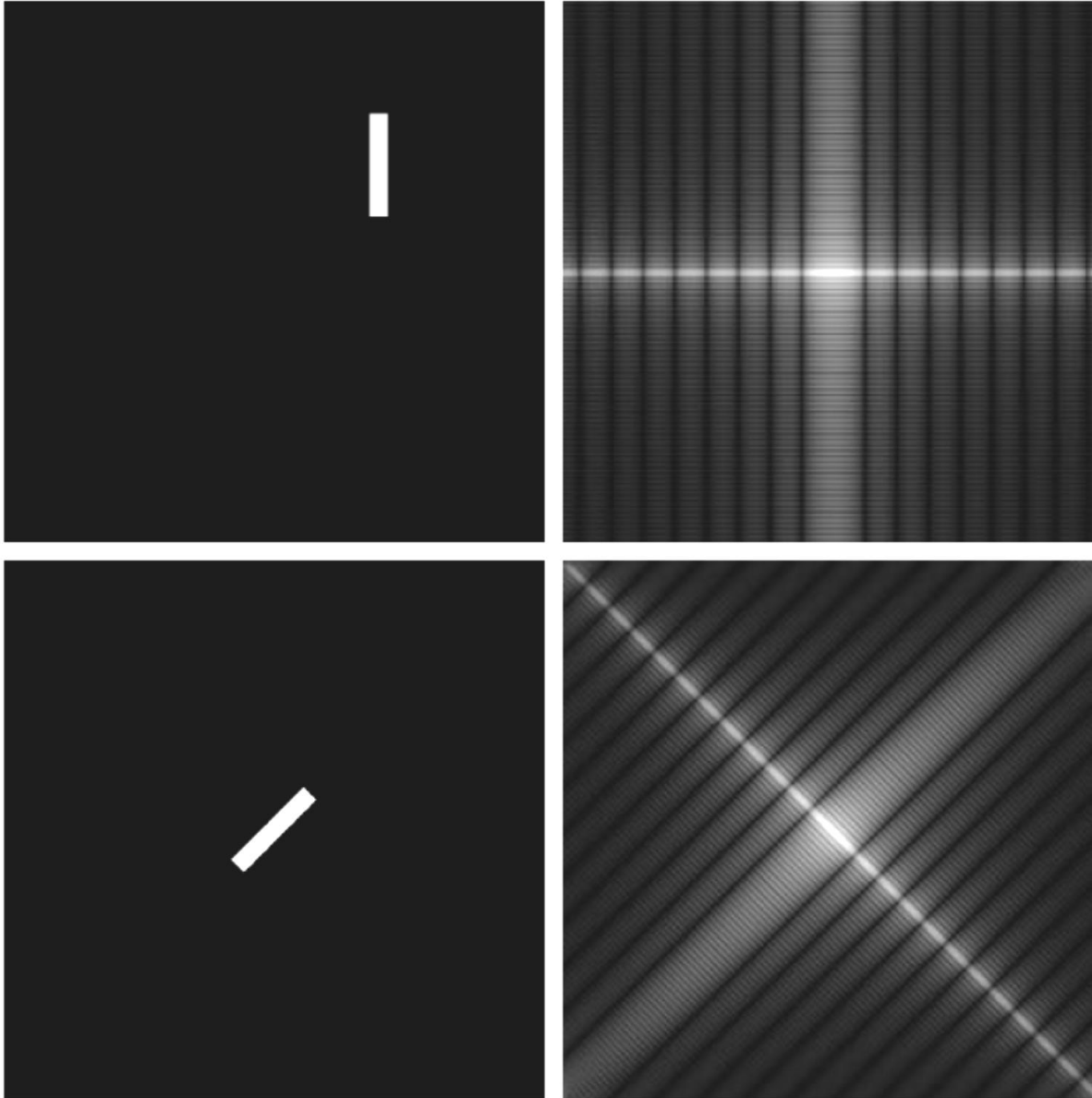
# Log Transformations

The energy of a spectrum is usually highly concentrated at the low-frequency region. A very strong contrast enhancement method, such as log transformation, is usually used to show details of the spectrum.

$$s = c \log(1 + r)$$



# 2-D Spectrum - Example



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Note the effect of strong intensity variations along particular directions:

