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# Sampling Theory

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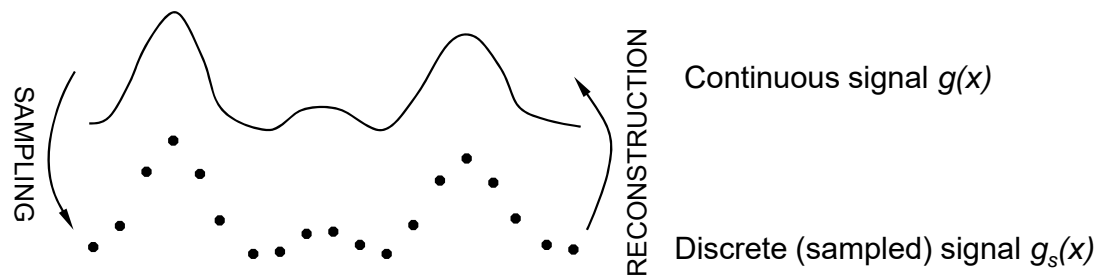
# Objectives

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- In this lecture we describe sampling theory:
  - Motivation
  - Analysis in the spatial and frequency domains
  - Ideal and nonideal solutions
  - Aliasing and antialiasing
  - Example: oversampling CD players

# Sampling Theory

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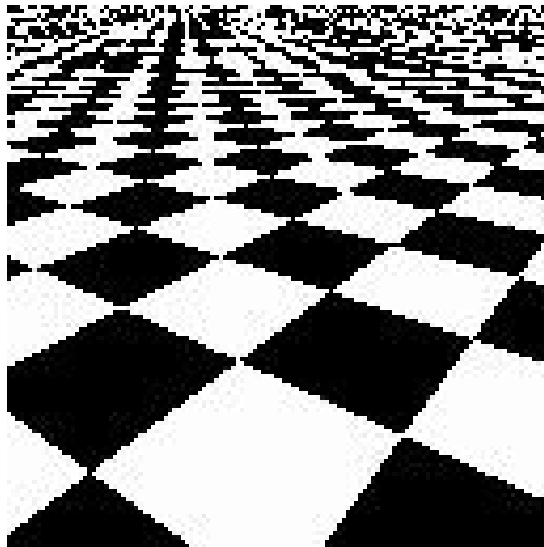
Sampling theory addresses the two following problems:

1. Are the samples of  $g_s(x)$  sufficient to exactly describe  $g(x)$ ?
2. If so, how can  $g(x)$  be reconstructed from  $g_s(x)$ ?

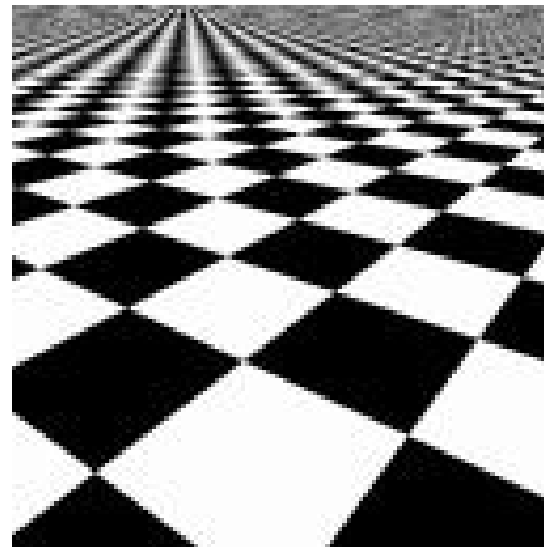
# Motivation

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- Improper consideration leads to jagged edges in magnification and Moire effects in minification.

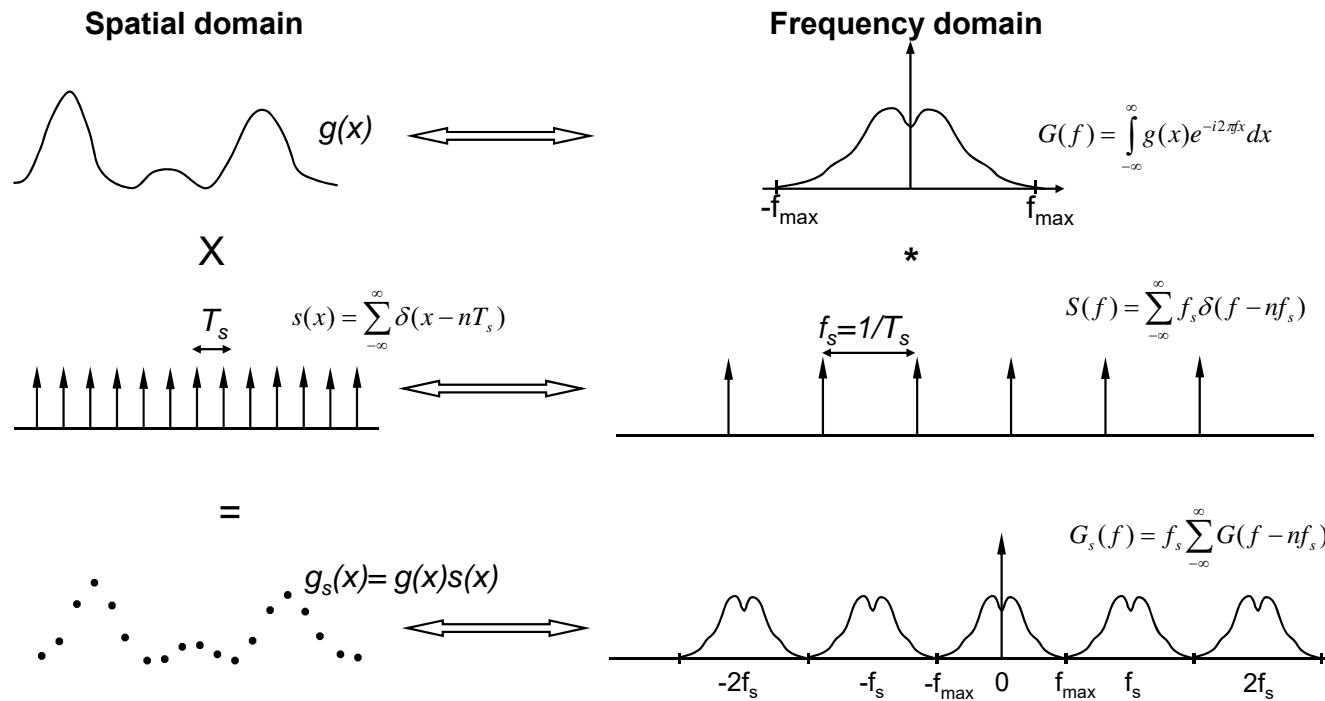


**Unfiltered image**



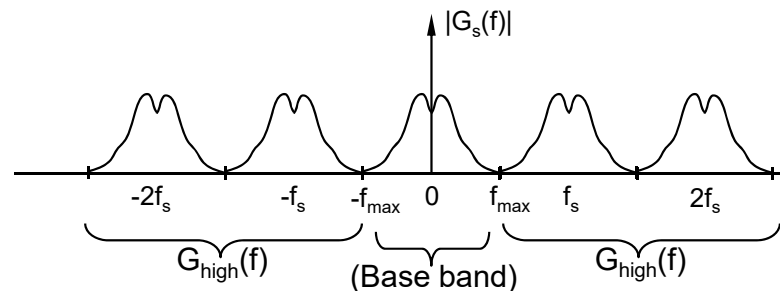
**Filtered image**

# Insight: Analysis in Frequency Domain



# Exploit Well-Known Properties of Fourier Transforms

1. Multiplication in spatial domain  $\leftrightarrow$  convolution in frequency domain.
2. Fourier transform of an impulse train is itself an impulse train.
3.  $G_s(f)$  is the original spectrum  $G(f)$  replicated in the frequency domain with period  $f_s$ .



$$G_s(f) = G(f) + G_{\text{high}}(f) \rightarrow \text{introduced by sampling}$$

# Solution to Reconstruction

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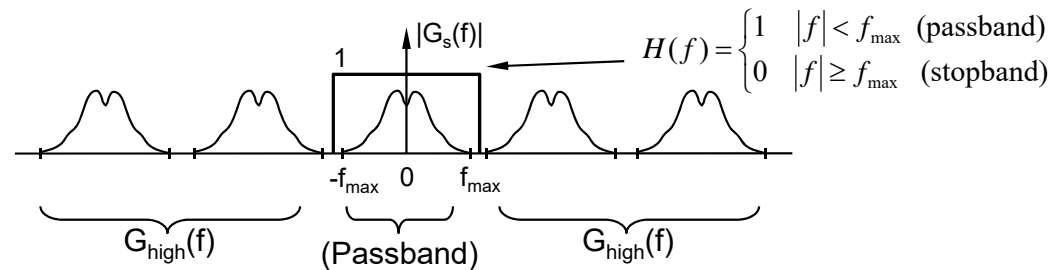
Discard replicas  $G_{high}(f)$ , leaving baseband  $G(f)$  intact.

Two Provisions for exact reconstruction:

- The signal must be bandlimited. Otherwise, the replicas would overlap with the baseband spectrum.
- $f_s > 2f_{max}$  (Nyquist frequency)

# Ideal Filter in the Frequency Domain: Box Filter

- Assuming provisions hold, we can discard  $G_{\text{high}}$  by multiplying  $G_s(f)$  with a box filter in the frequency domain.



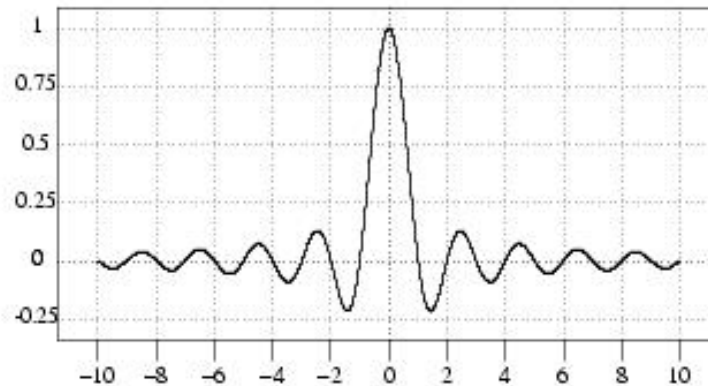


# Ideal Filter in the Spatial Domain: Sinc Function

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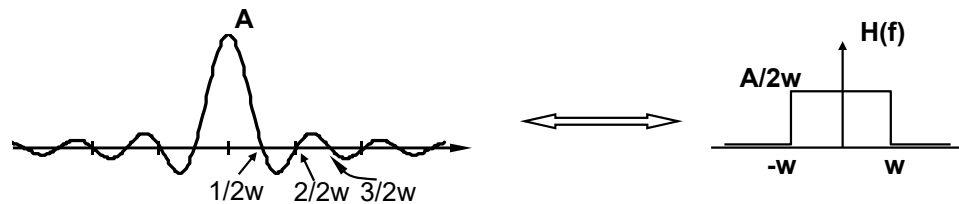
The ideal lowpass filter in the spatial domain is the inverse Fourier transform of the ideal box:

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



# Reciprocal Relationship

- Reciprocal relationship between spatial and frequency domains:



- For interpolation, A must equal 1 (unity gain when sinc is centered at samples; pass through 0 at all other samples)

$W = f_{\max} = 0.5$  cycle / pixel for all digital images

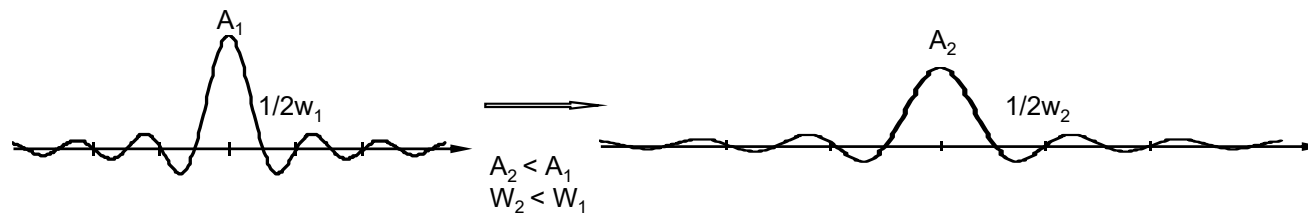


Highest freq: on-off sequence

1 cycle is 2 pixels (black, white, black, white), therefore  $\frac{1}{2}$  cycle per pixel is  $f_{\max}$ .

# Blurring

- To blur an image,  $W \downarrow$  and  $A \downarrow$  so that  $A/2W$  remains at 1.



- To interpolate among sparser samples,  $A=1$  and  $W \downarrow$  which means that  $(A/2W) \uparrow$ . Since sampling decreases the amplitude of the spectrum,  $(A/2W) \uparrow$  serves to restore it.

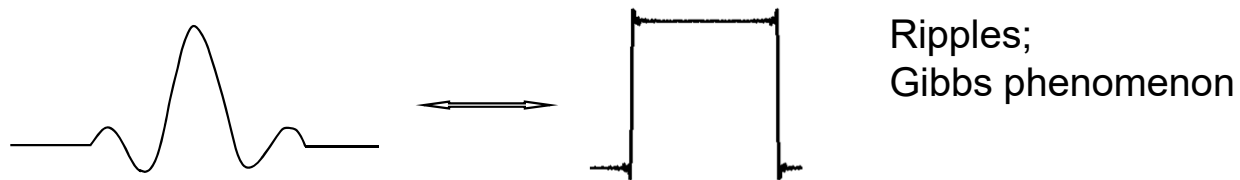
$$g(x) = \text{sinc}(x) * g_s(x)$$
$$= \int_{-\infty}^{\infty} \text{sinc}(\lambda) g_s(x - \lambda) d\lambda$$

*Note* : sinc requires infinite number of neighbours : impossible.

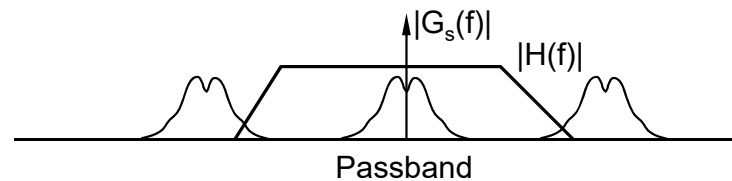
- Ideal low-pass filtering is impossible in the spatial domain. It is not impossible in the frequency domain for finite data streams.

# Nonideal Reconstruction

- Possible solution to reconstruction in the spatial domain: truncated sinc.



- Alternate solution: nonideal reconstruction



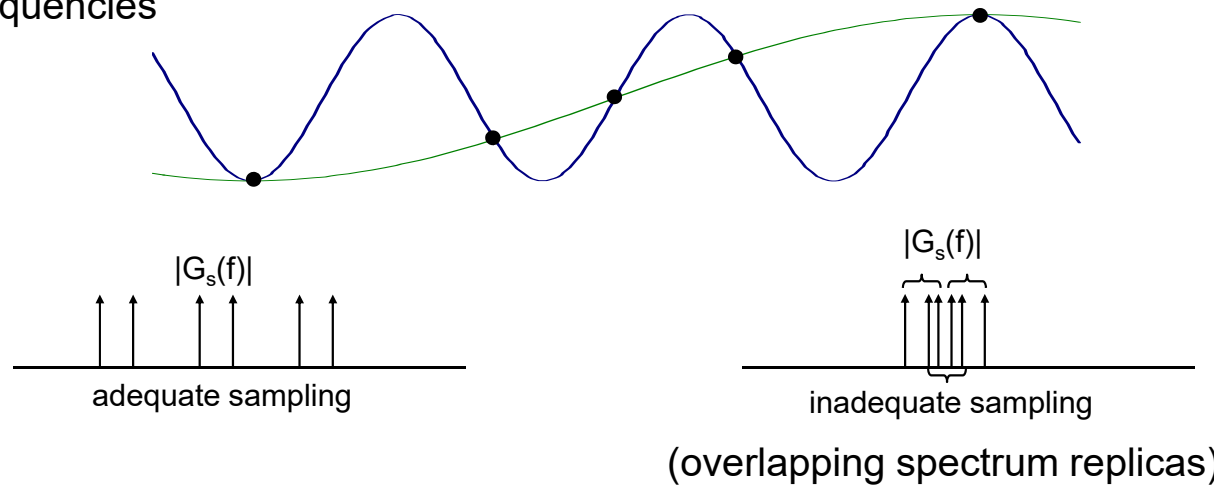
- Nonideal because it attenuates the higher frequencies in the passband and doesn't fully suppress the stopband.

# Aliasing

If  $f_s < 2f_{\max}$  our signal is undersampled.

If  $f_{\max}$  is infinite, our signal is not bandlimited.

Either way, we have aliasing: high frequencies masquerade, or alias as, low frequencies

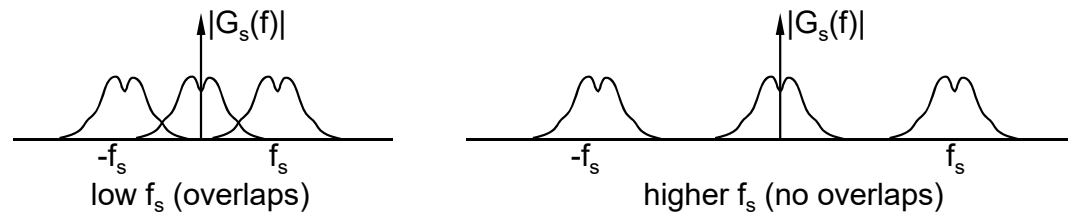


Temporal aliasing: stagecoach wheels go backwards in films  
(sampled at 24 frames/sec).

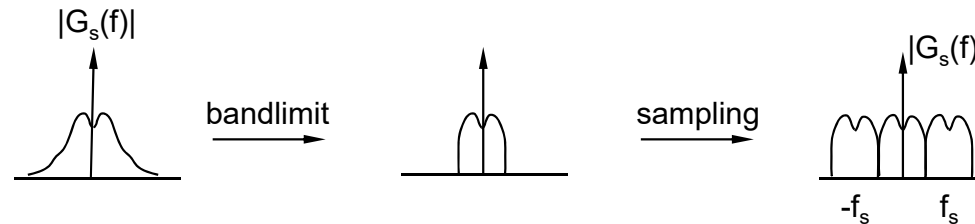
# Antialiasing

Two approaches to antialiasing for combating aliasing:

1. Sample input at higher rate: increase  $f_s$



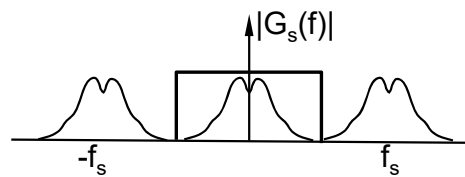
2. bandlimit input before sampling at low  $f_s$



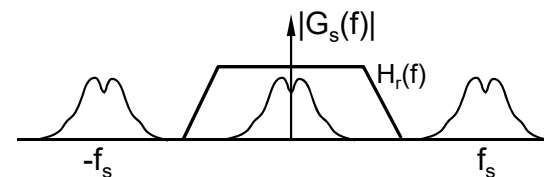
# Antialiasing

Intuition for increasing  $f_s$ :

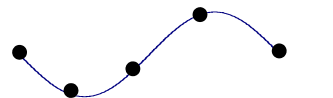
Higher sampling rates permit sloppier reconstruction filters.



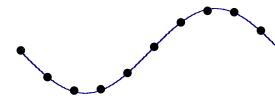
Marginally acceptable  $f_s$   
Requires ideal lowpass filter



high  $f_s$  allows nonideal reconstruction filter.  
This adequate because  $H_r(f)$  doesn't taper off until  $|f| > |F_{max}|$



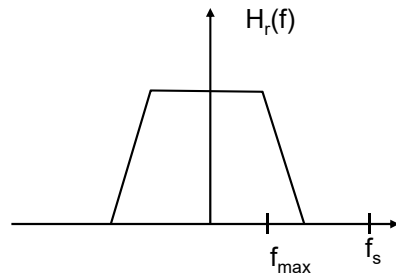
Pushed to limits; use sinc



Many samples: linear interpolation adequate

# Nonideal Reconstruction Filter

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$$H_r(f) = \begin{cases} 1 & 0 \leq |f| \leq f_{\max} \\ \neq 0 & f_{\max} < |f| < f_s - f_{\max} \\ 0 & |f| \geq f_s - f_{\max} \end{cases}$$



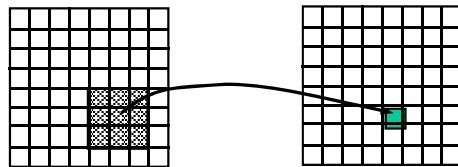
# Nonideal Reconstruction Filter

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Intuition for bandlimiting input:

Inability to change  $f_s$ : e.g., restricted by display resolution.

For example, an image in perspective may require infinite pixels to capture all visual details. However, only finite # pixels available, e.g.  $512 \times 512$

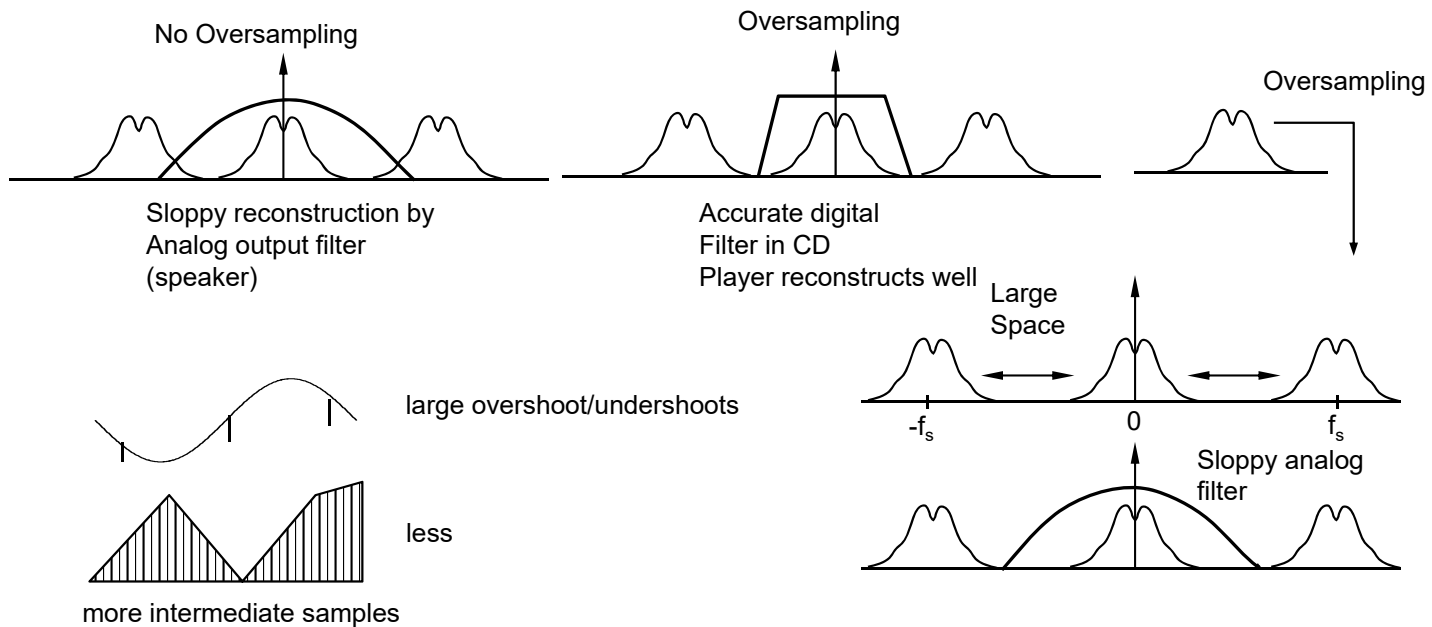


Many pixels in oblique image map to one output pixel. Must blur neighborhood before returning a single value to the output image. In general, dull signals don't need bandlimiting, rich ones do.

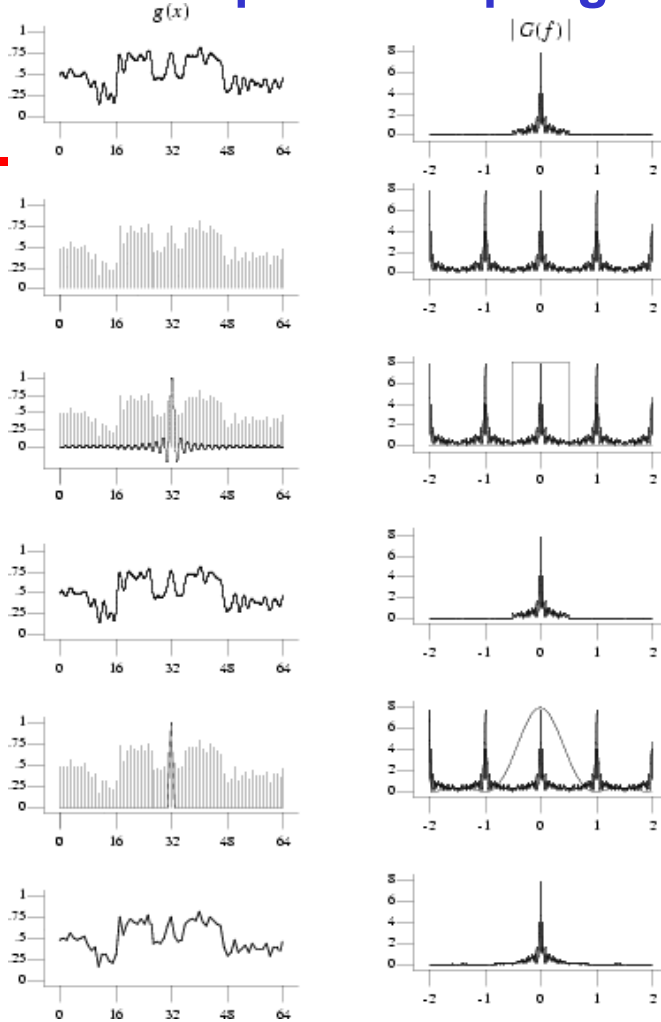
# Example: Oversampling CD players

$$f_s = 44.1 \text{ kHz (sample/second)}$$

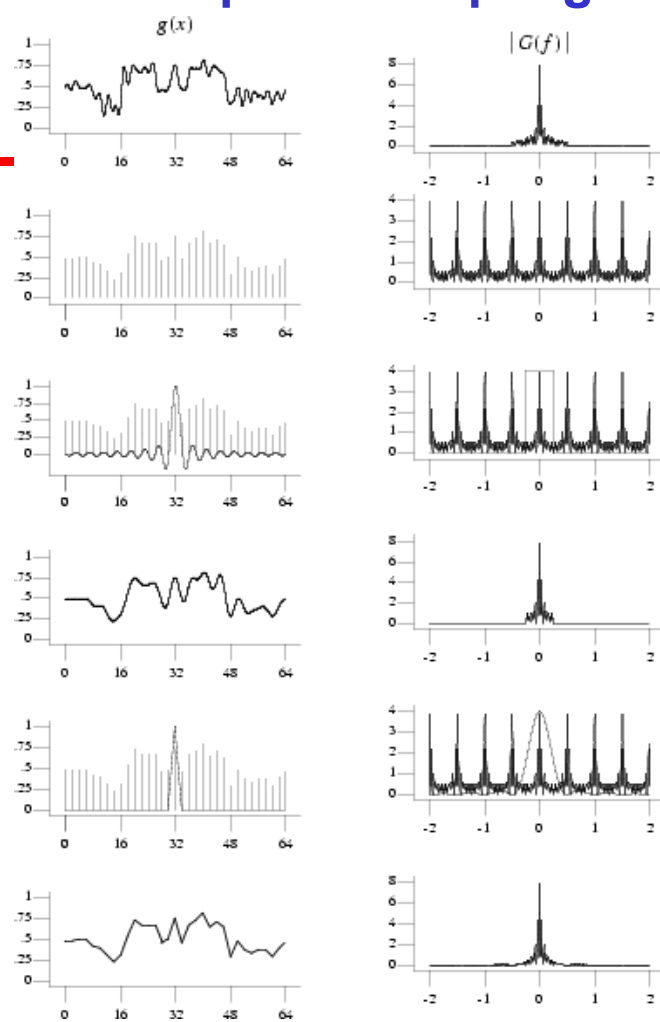
$$= 20 \text{ kHz} \underset{f_{max}}{*} 2 \underset{\text{Nyq rate}}{+} 4.1 \text{ kHz} \underset{\text{cushion}}{=} 44.1 \text{ kHz}$$



## Adequate Sampling



## Inadequate Sampling



# Antialiasing

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