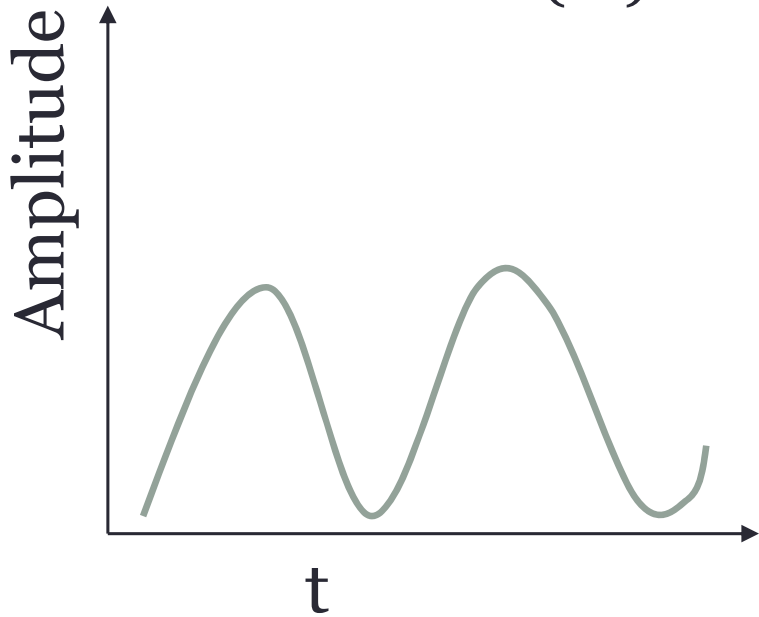


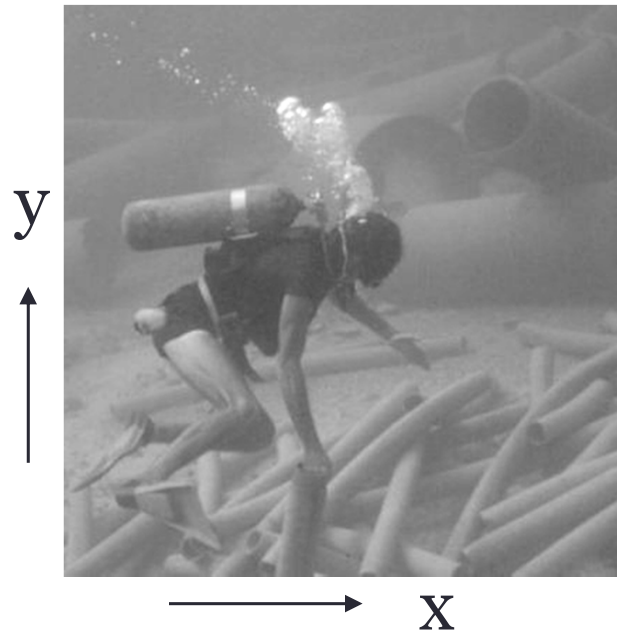
What is a signal?

- Any function dependent on a single or multiple variables

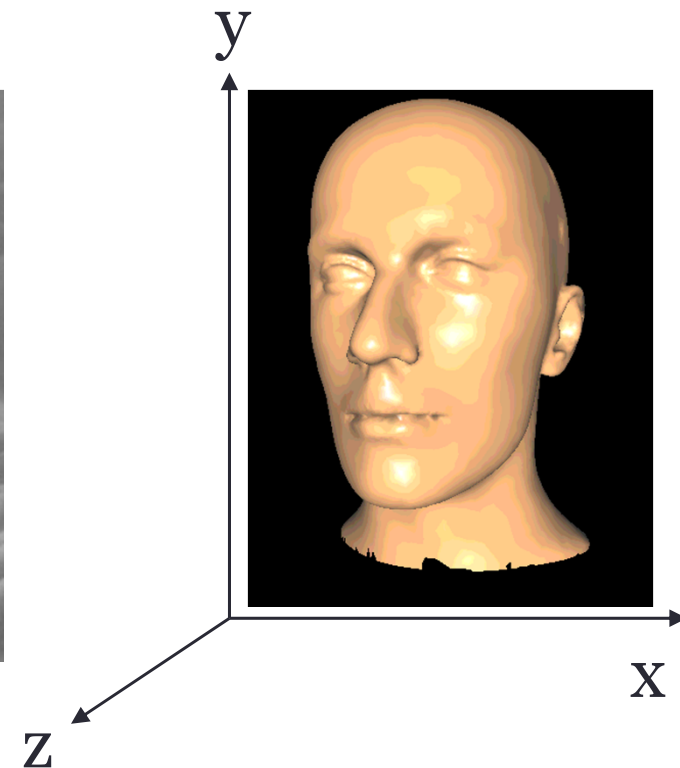
$$1D: A = f(t)$$



$$2D: I = f(x, y)$$



$$3D: M = f(x, y, z)$$



Origin of Signals

- Real life applications
 - Medical Images
 - Seismic Vibrations
 - Audio and Video
 - Radar and Sonar

Signal Processing

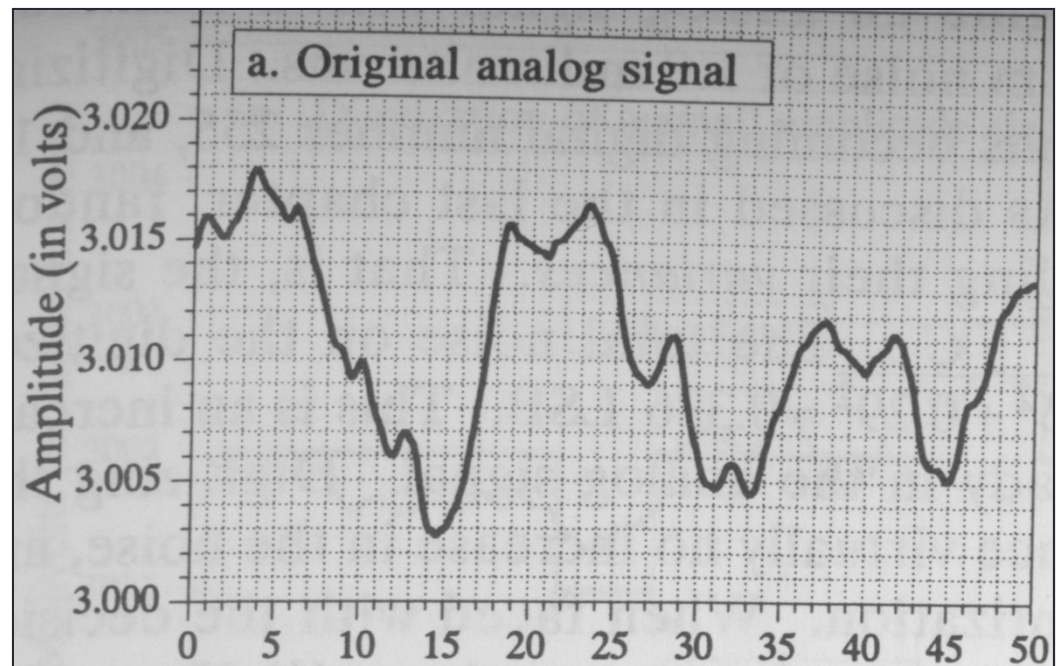
- Science of processing signals
 - Data analysis
 - Data compression
 - Data Storage
 - Data Retrieval
 - Data Communication
 - Special Effects

Applications

- Space
 - Photo Enhancement, Segmentation
- Medical
 - Diagnostic image analysis
- Media
 - Audio and video compression, Quality Control
- Military
 - Radar and sonar data analysis
- Scientific
 - Emergency monitoring and analysis

Analog Signals

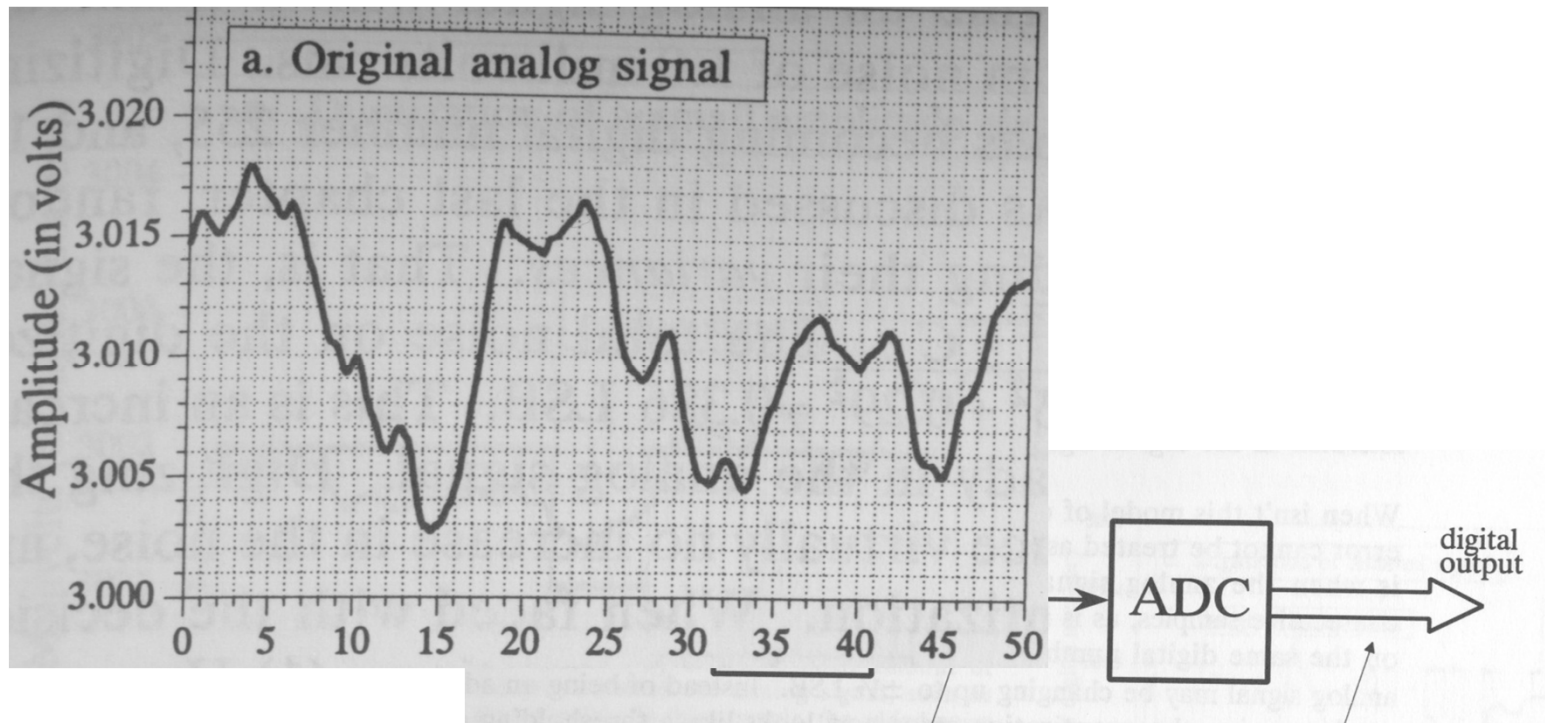
- Both independent and dependent variables can assume a continuous range of values
- Exists in nature



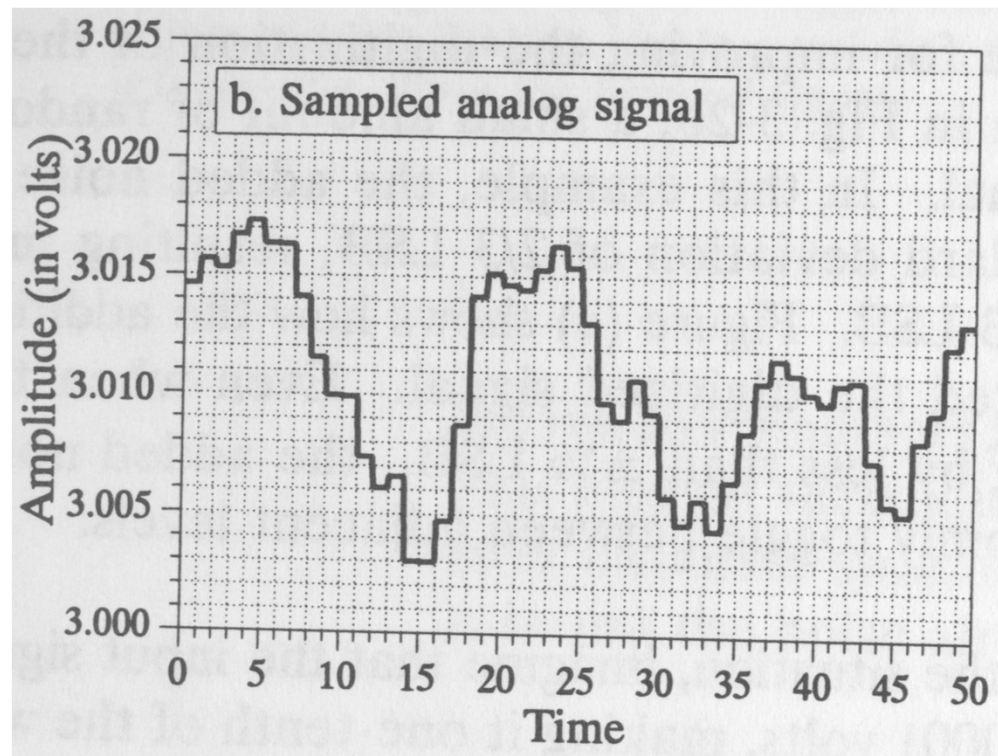
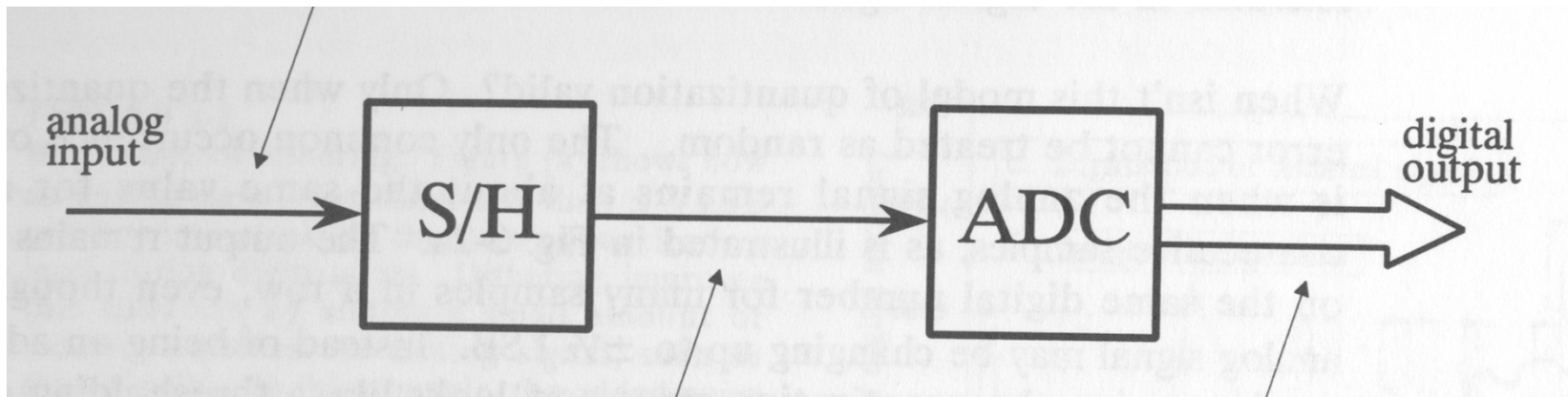
Digital Signals

- Both independent and dependent variables are discretized
- Representation in computers
- *Sampling*
 - Discrete independent variable
 - Sample and hold (S/H)
- *Quantization*
 - Discrete dependent variable
 - Analog to Digital Converter (ADC)

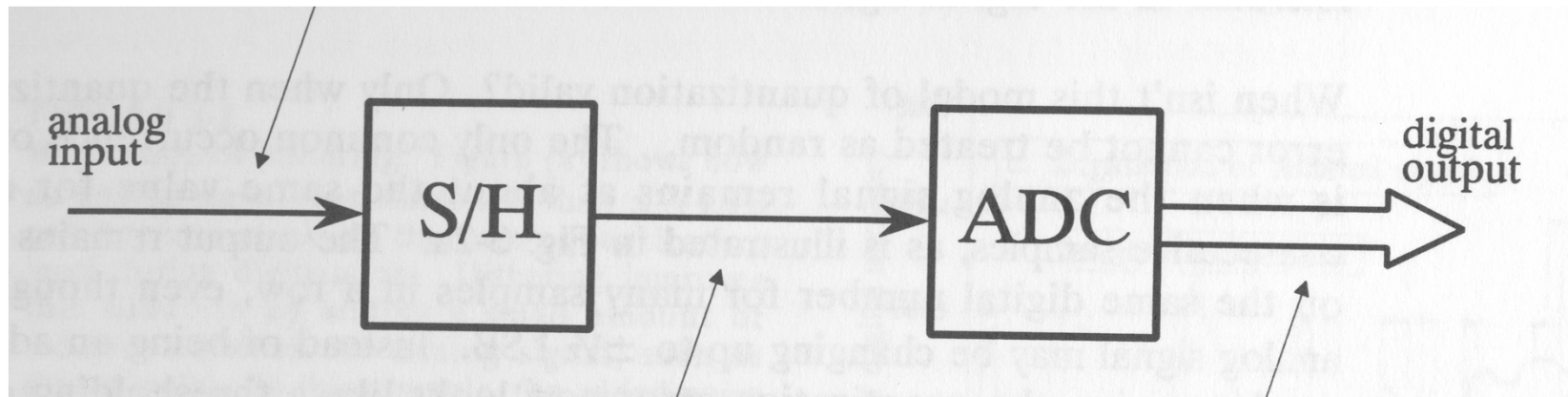
Digital Signal



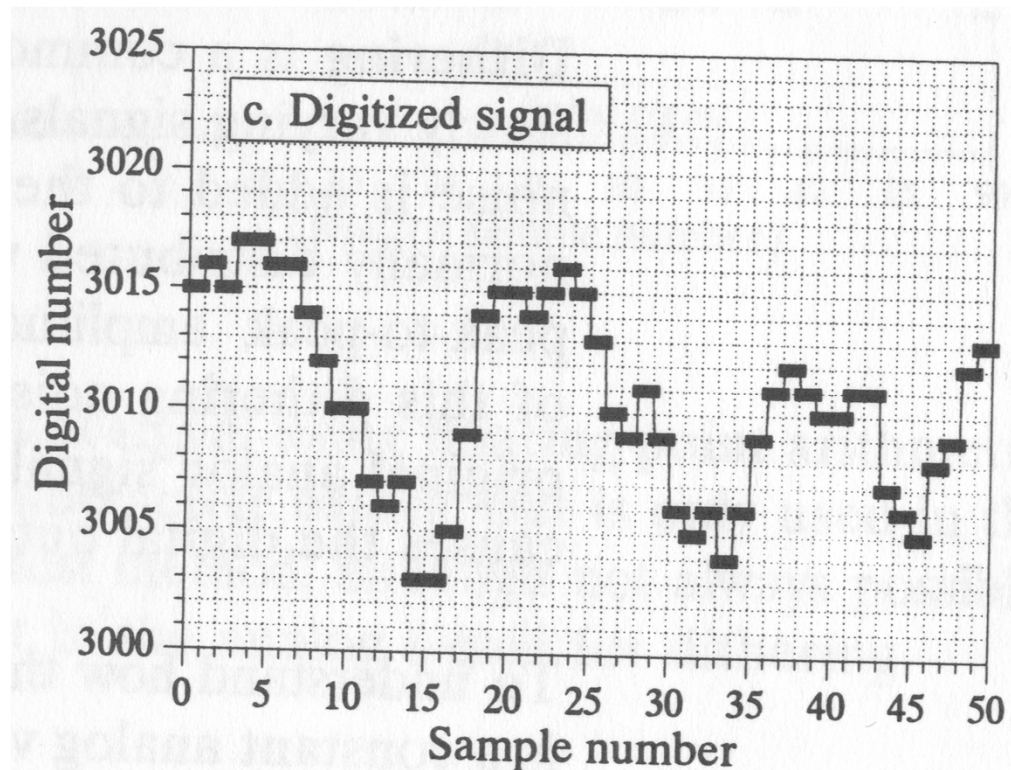
Sampled Signal



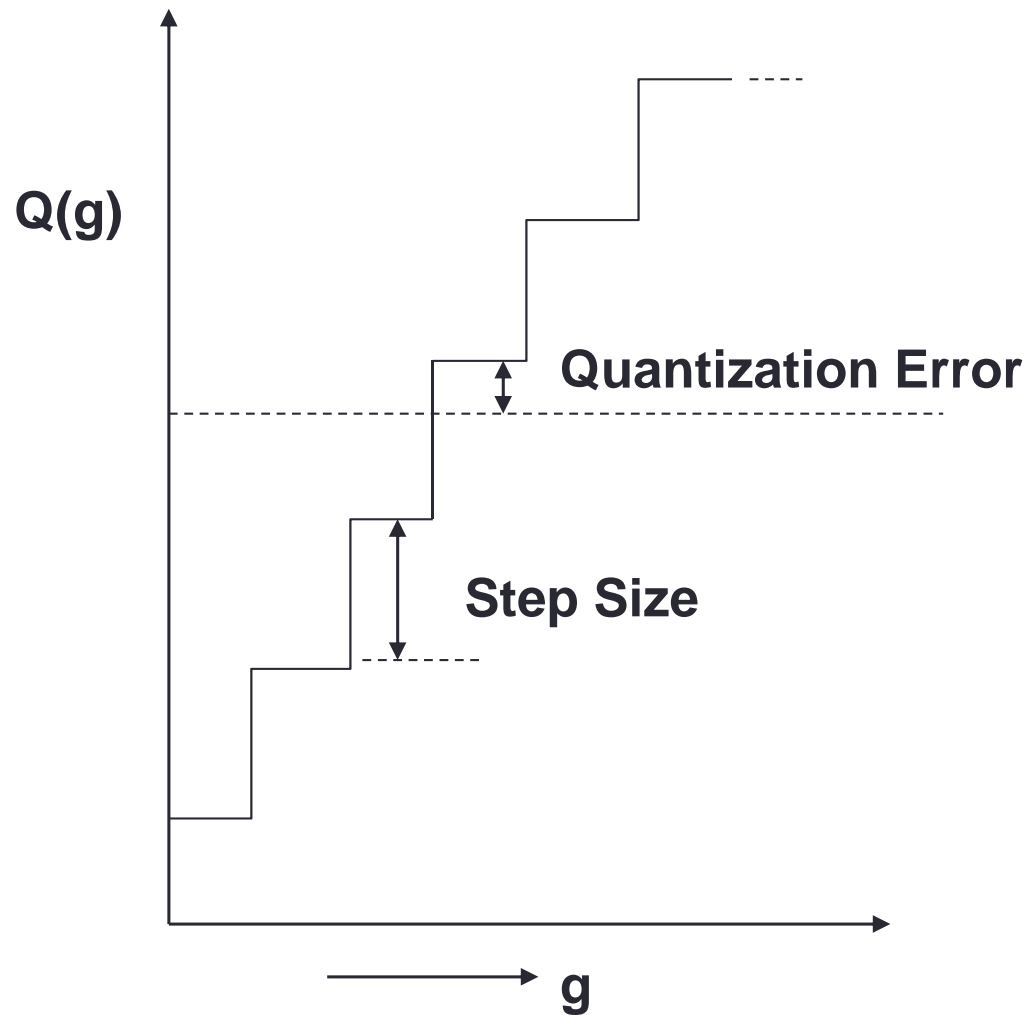
Digitized Signal



- Depends on number of bits
- 12 bits = 4095 levels
- $0.0 \leq \text{Voltage} \leq 4.096$
- 2.56 and 2.5601 TO 2560
- Each level (LSB) = 0.001
- Error $\leq \pm 1/2$ LSB
- Called *Quantization Error*



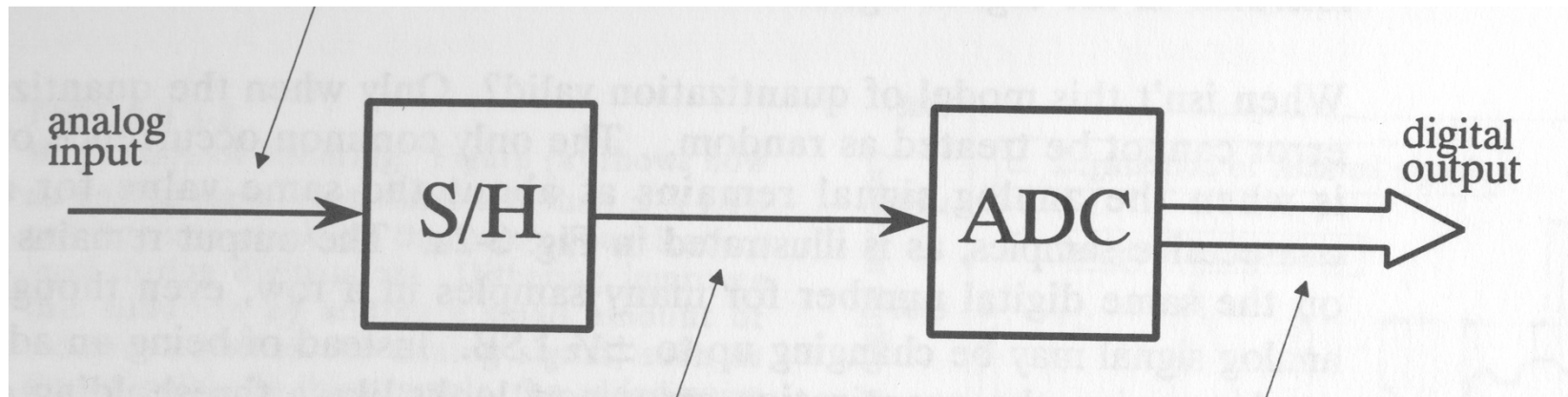
Quantization Error



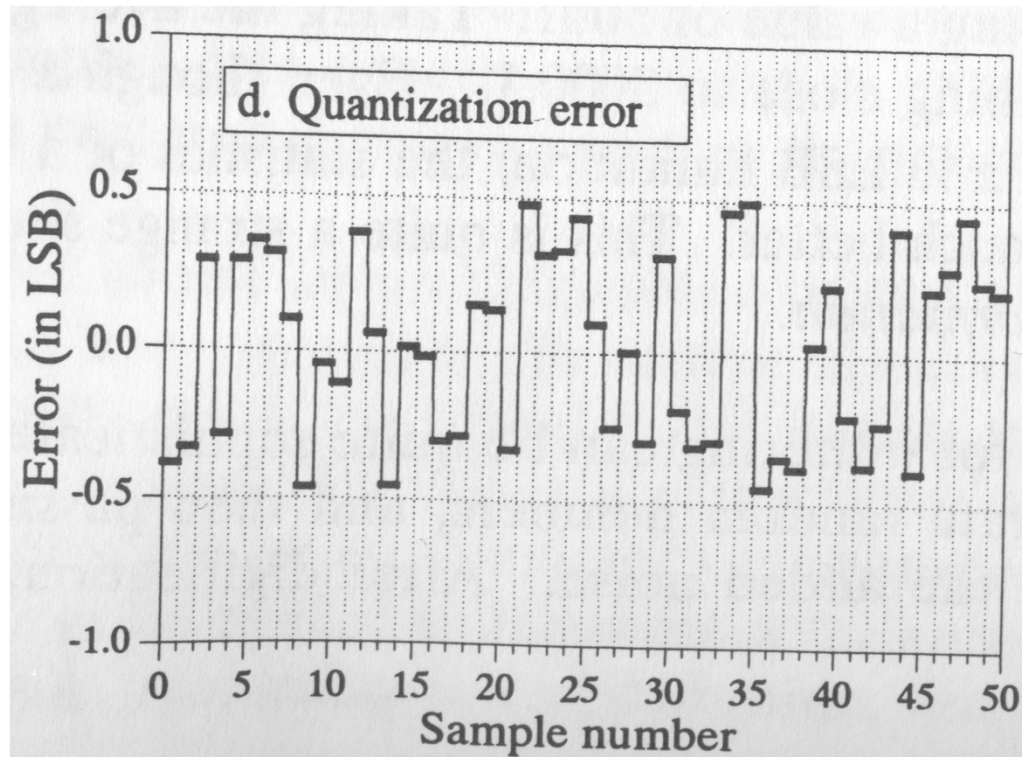
Uniform Quantization

Maximum Error = $\frac{1}{2}$ Step Size

Digitized Signal

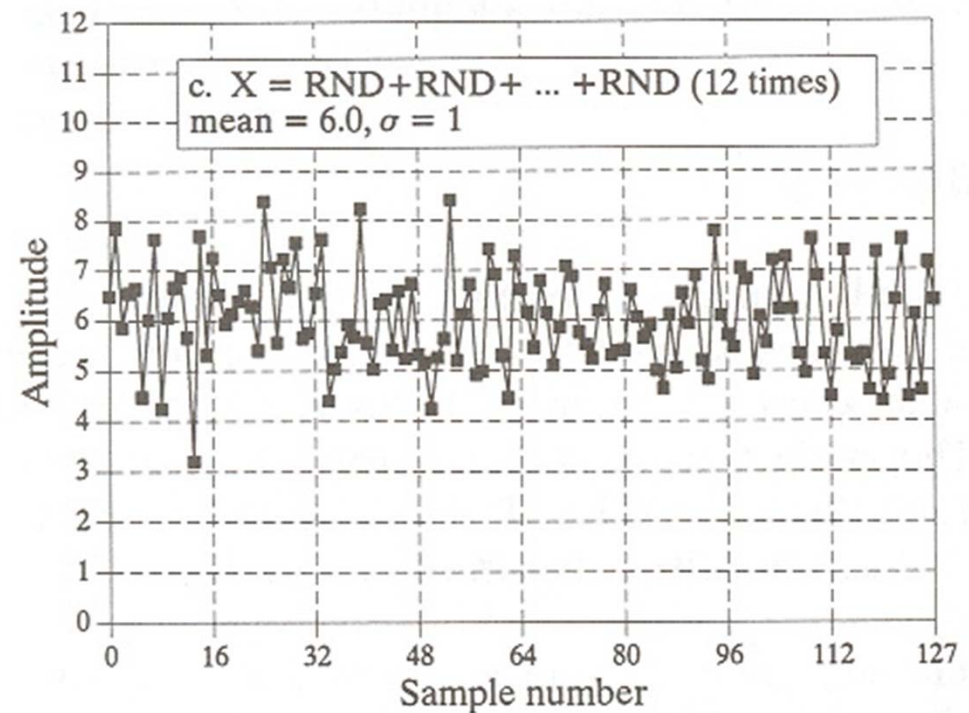


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Quantization Error

- Usually like random noise
- Noise is present in most signal acquisition systems
- Random uncorrelated samples added to the original signal



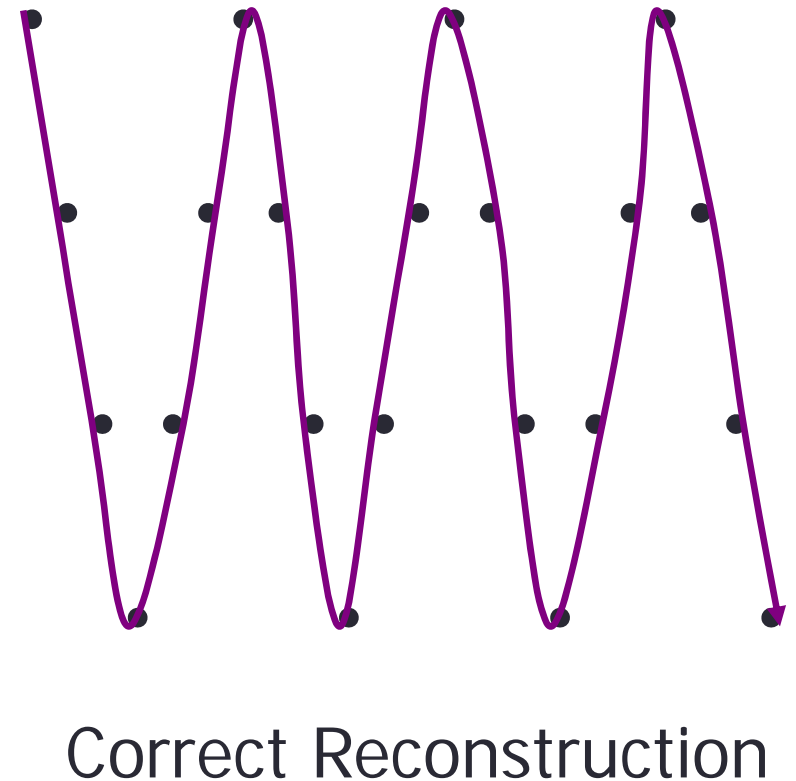
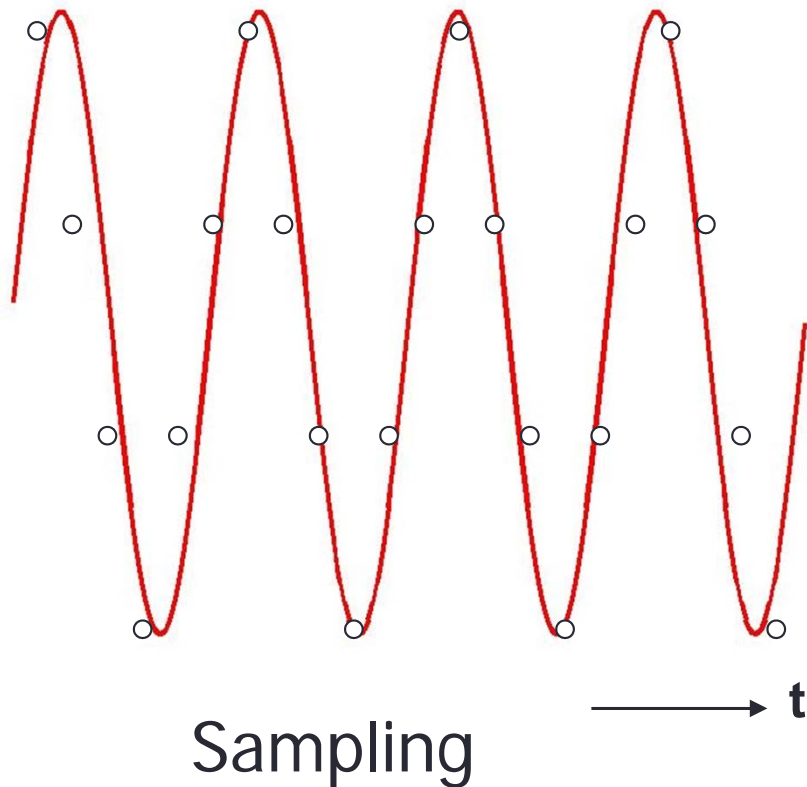
Proper Sampling

- If the original signal can be reconstructed *unambiguously* from the sampled signal

$$\begin{aligned}\text{Cycles/ Sample} &= \frac{\text{Number of cycles per second}}{\text{Number of samples per second}} \\ &= \frac{\text{Analog Frequency}}{\text{Sampling Rate}}\end{aligned}$$

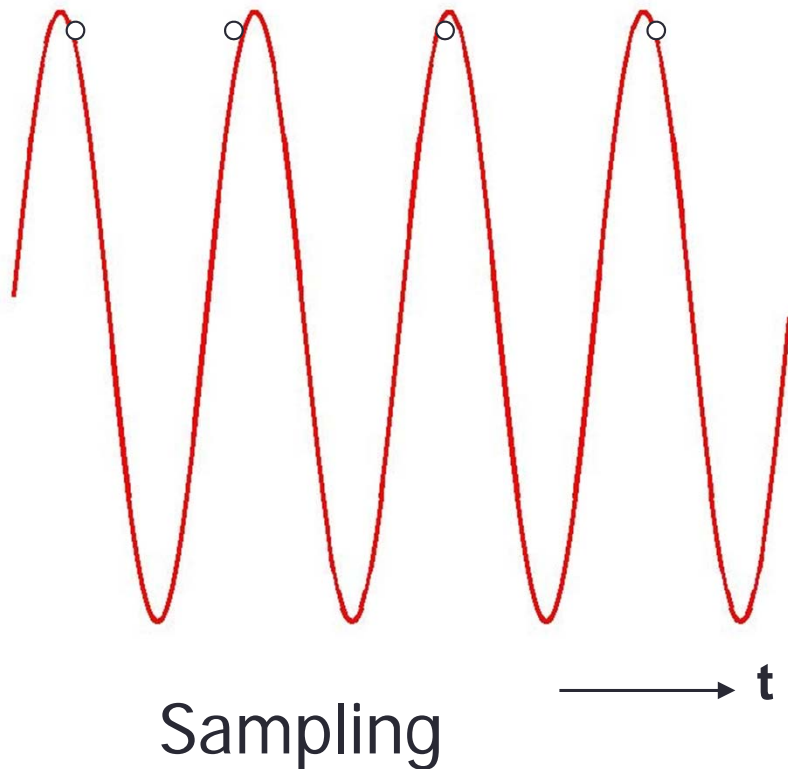
Digital Signals

- Defined at only few values of t



Digital Signals

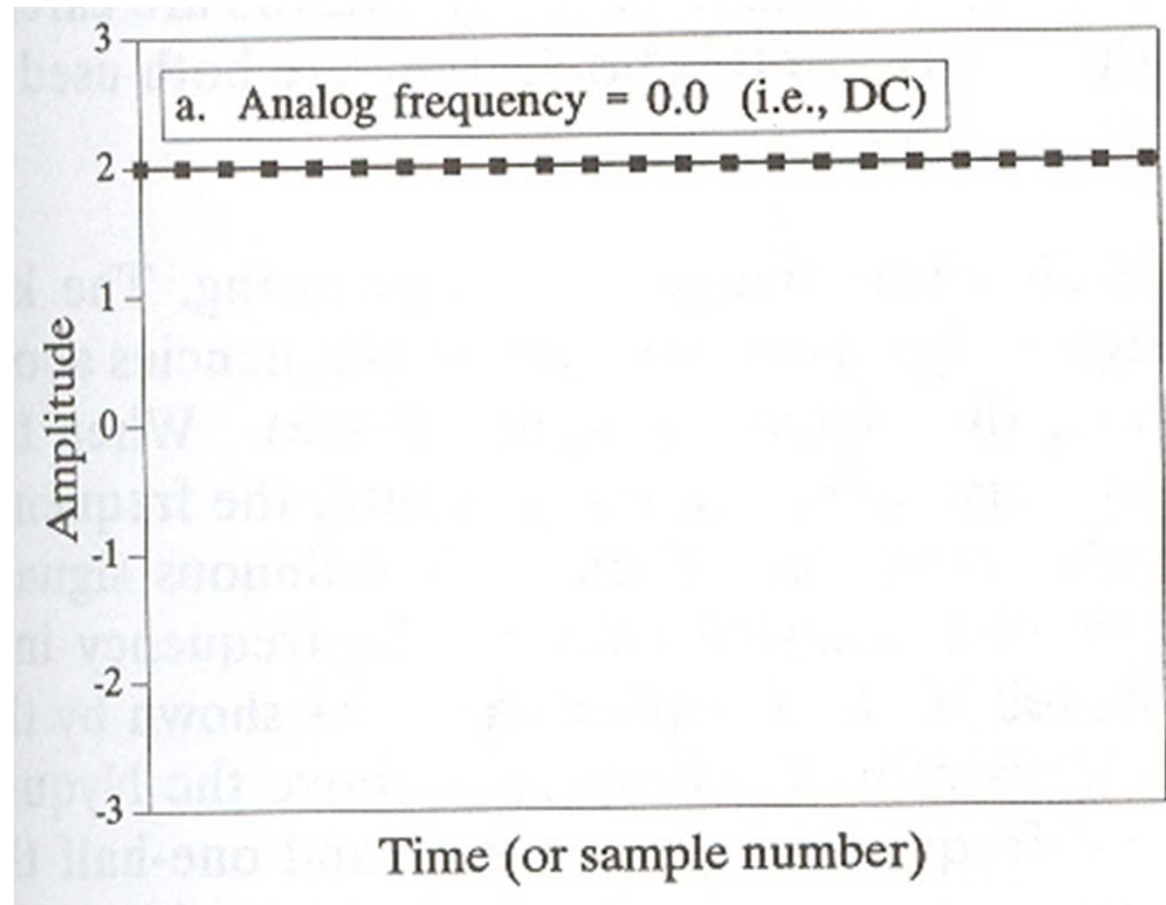
- Whether you can reconstruct correctly depends on how you sample – sampling rate



Incorrect Reconstruction

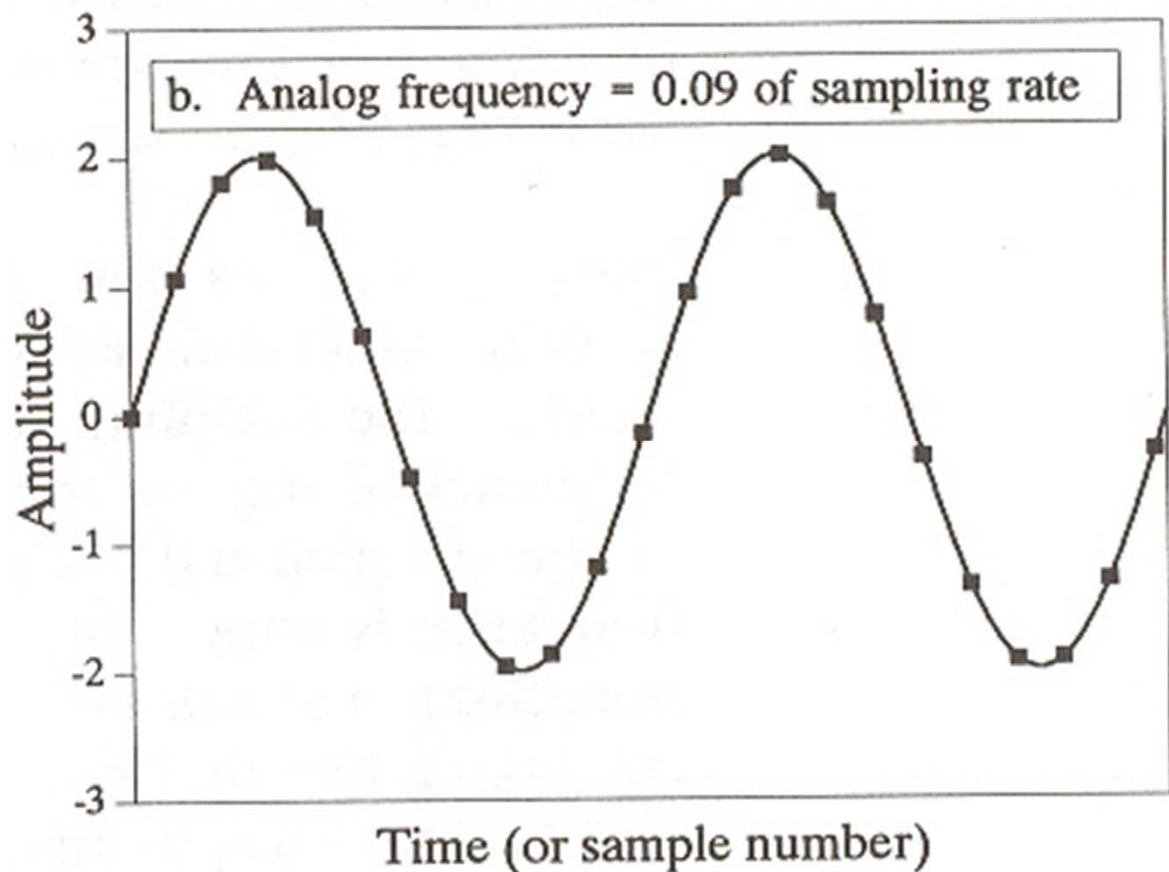
Is it Proper Sampling?

- DC signal
- $\text{Freq} = 0.0 \times \text{Sampling Rate}$
- Proper



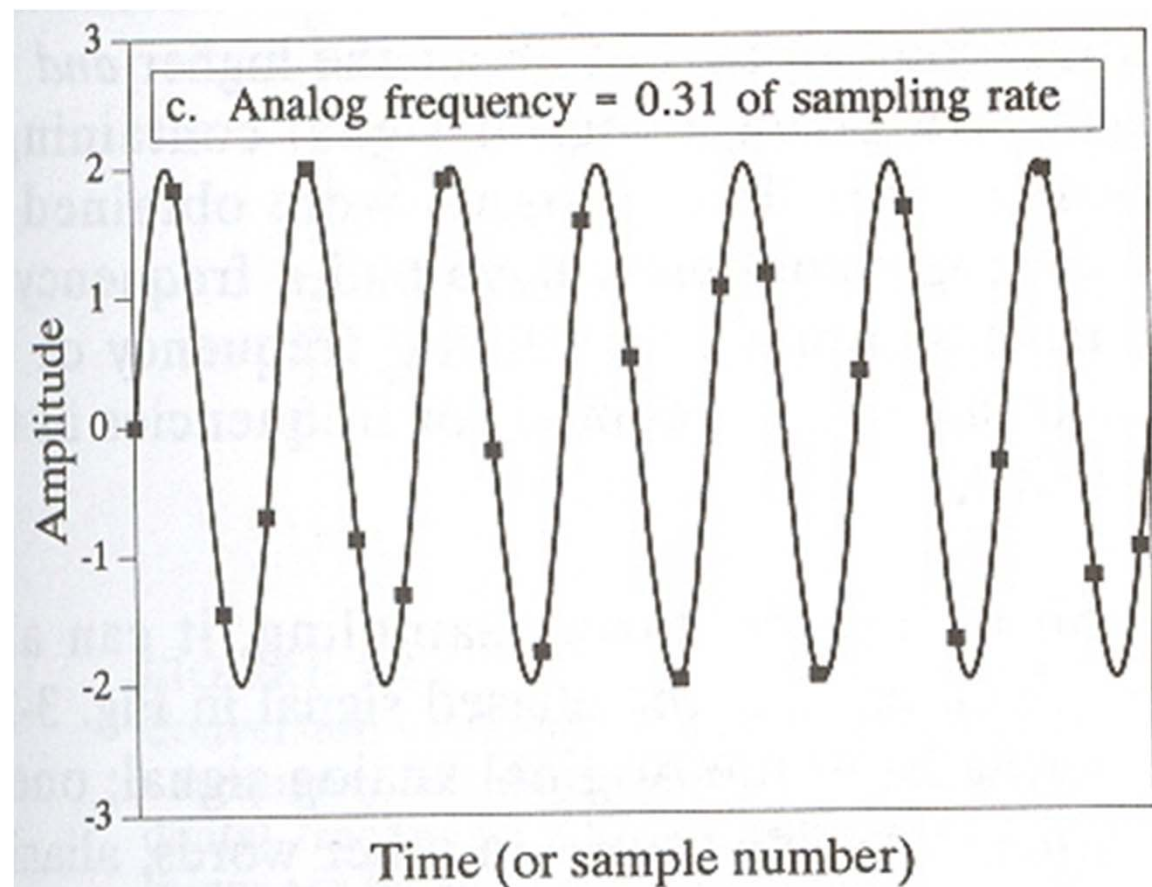
Is it Proper Sampling?

- $\text{Freq} = 0.09 \times \text{Sampling Rate}$
- Each sample covers 0.09 cycles
- Proper



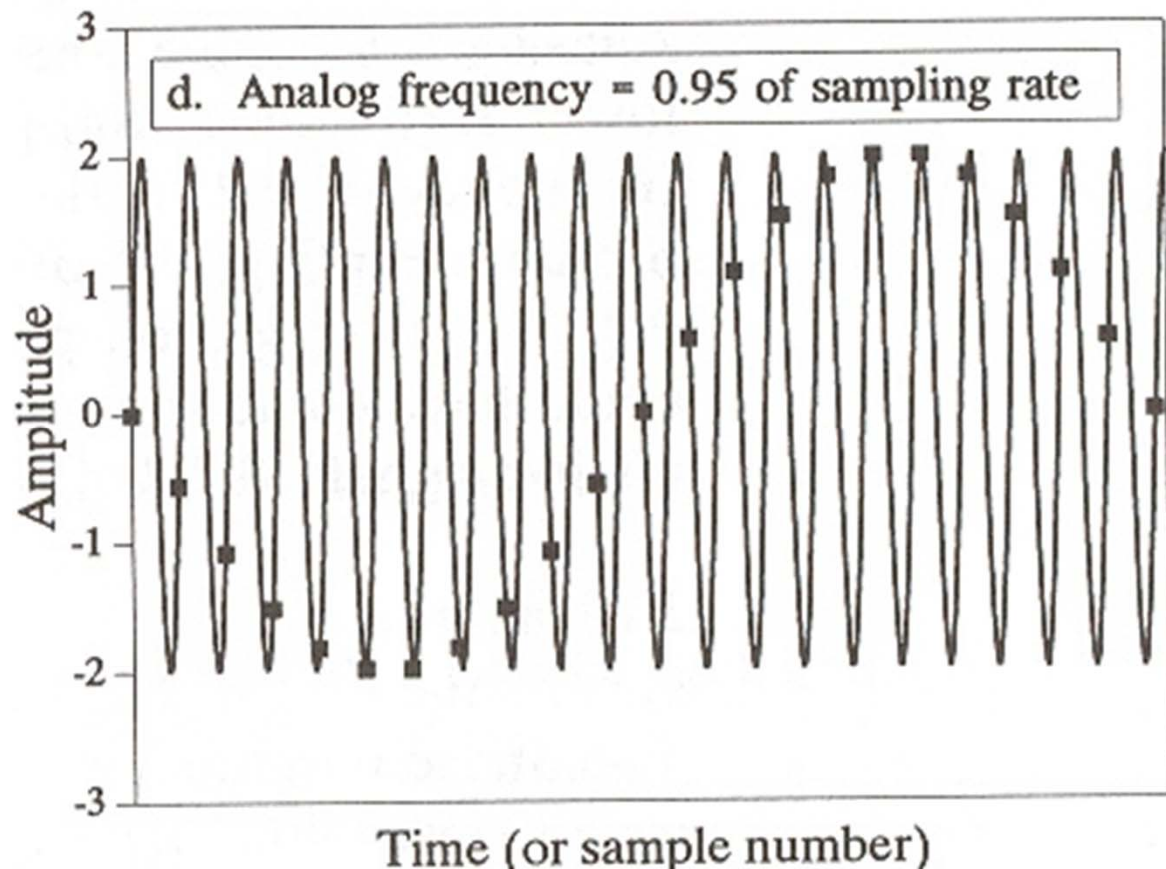
Is it Proper Sampling?

- $\text{Freq} = 0.31 \times \text{Sampling Rate}$
- Larger fraction of cycles per sample
- Proper



Is it Proper Sampling?

- $\text{Freq} = 0.95 \times \text{Sampling Rate}$
- Much larger parts of cycles per sample
- Not Proper
- Aliasing
- Changes frequency and phase



Sampling Theorem

- Proper Sampling: At least one sample per half cycle
- $\text{Freq} \leq 0.5 \times \text{Sampling Rate}$
- $\text{Sampling Rate} \geq 2 \times \text{Frequency}$
- Nyquist Rate