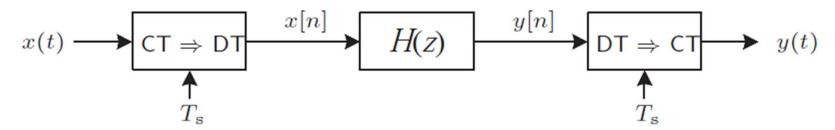
## Digital Signal Processing (DSP)

Fall 2014
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#### DIGITAL SIGNAL PROCESSING (DSP)

# Lecture 7 Sampling Theorem

DT processing of CT signals

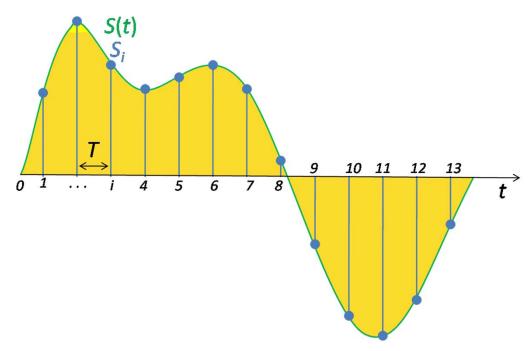


 We work on discrete-time signals not digital signals (more on this issue later!)

http://web.cecs.pdx.edu/~ece2xx/ECE223/Slides/Sampling.pdf

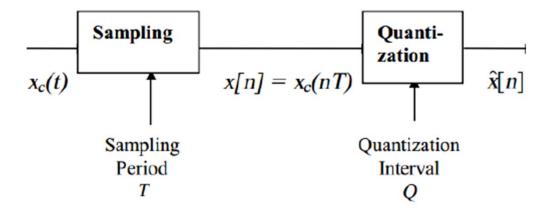
• The main step: sampling  $x[n] = x(t)|_{t=nT_s}$ 

$$x[n] = x(t)|_{t=nT}$$

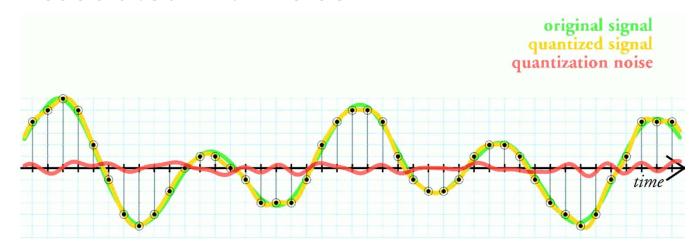


http://en.wikipedia.org/wiki/Sampling %28signal processing%29

- Sampling: conversion of CT signals to discretetime signals
- Quantization: conversion of the discrete-time signals to discrete-time discrete-values signals

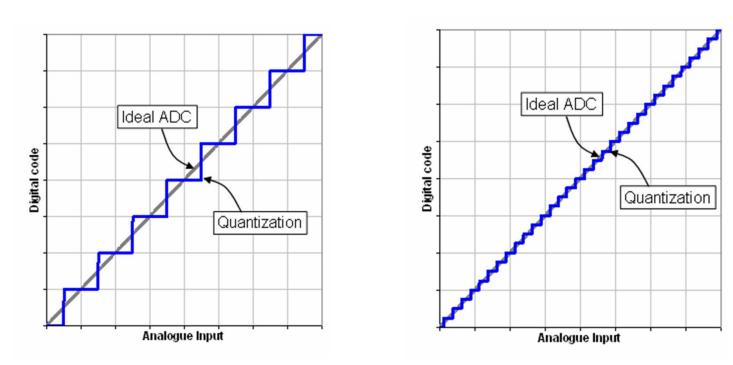


- Quantization: quantization level
  - Non-linear
  - Associated with noise



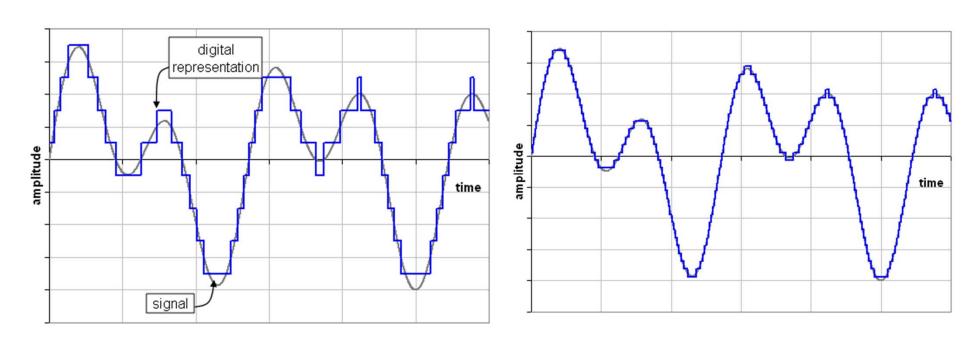
http://en.wikipedia.org/wiki/Quantization\_%28signal\_processing%29

More on the effects of the quantization level



http://www.diracdelta.co.uk/science/source/q/u/quantization%20error/source.html#.VHROe8nzxtA

#### Example

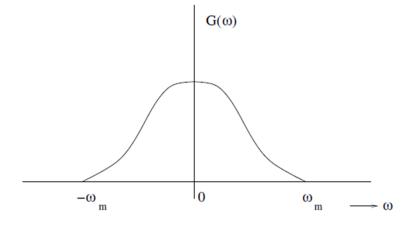


http://www.diracdelta.co.uk/science/source/q/u/quantization%20error/source.html#.VHROe8nzxtA

Nyquest theorem (sampling theorem)

"A bandlimited signal can be reconstructed exactly if it is sampled at a rate atleast twice the maximum frequency component in it."

Illustration: band-limited signal



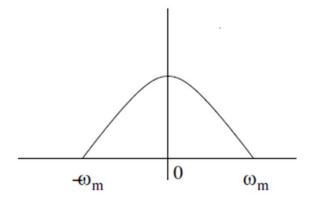
#### • Note that:

The maximum frequency component of g(t) is  $f_m$ . To recover the signal g(t) exactly from its samples it has to be sampled at a rate  $f_s \geq 2f_m$ .

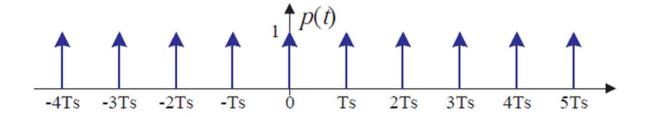
- Nyquest rate: the minimum required sampling frequency (Hz)
- Sampling frequency (Hz)

- Example: speech signal with 4 KHz bandwidth
  - Nyquest raet: 8000Hz
  - Sampling frequency: more than 8000Hz
  - Usual quantization level: 256
- ADC: sampling & quantization
  - Sampling frequency
  - Quantization level
  - Challenge!

- Important proof:
  - Consider the band-limited signal x(t)
  - We use a modeling for the sampled signal
  - We calculate the FT to observe the proof



Impulse sampling

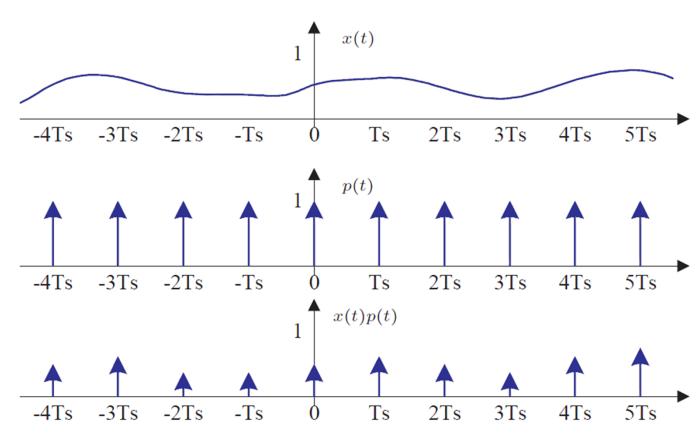


$$p(t) = \sum_{n = -\infty}^{\infty} \delta(t - nT_{\rm s})$$

We model the sampled signal by weighted impulses

http://web.cecs.pdx.edu/~ece2xx/ECE223/Slides/Sampling.pdf

#### Example



#### Define

$$x_{\delta}(t) = x(t) p(t) = \sum_{n=-\infty}^{\infty} x(nT_{\rm s})\delta(t - nT_{\rm s})$$

Next we compute the CTFT

$$x(t) p(t) \stackrel{\mathcal{F}T}{\iff} \frac{1}{2\pi} X(j\omega) * P(j\omega)$$

– Remember that (lecture 6):

$$P(j\omega) \stackrel{\mathcal{F}T}{\iff} \frac{2\pi}{T_{\rm s}} \sum_{k=-\infty}^{\infty} \delta\left(\omega - k\frac{2\pi}{T_{\rm s}}\right)$$

Therefore,

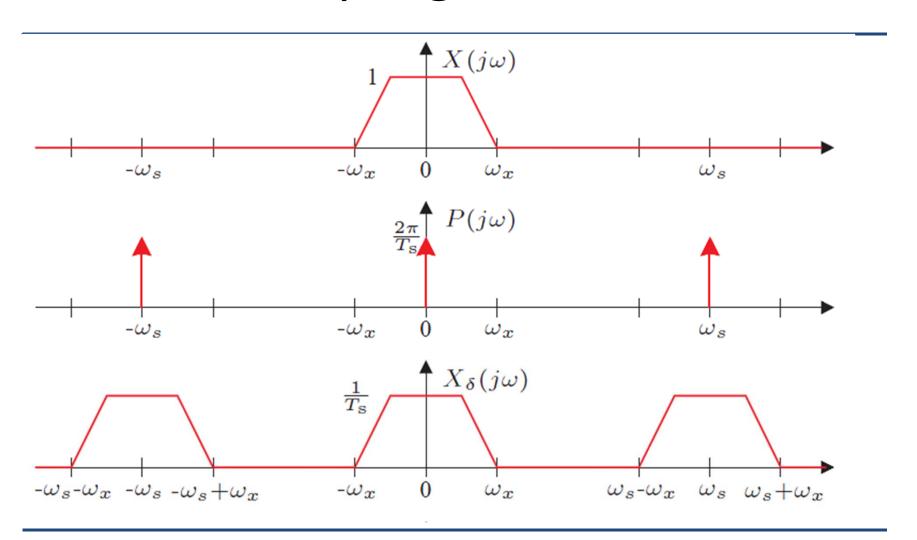
$$x(t) p(t) \stackrel{\mathcal{F}T}{\iff} \frac{1}{2\pi} X(j\omega) * \frac{2\pi}{T_s} \sum_{k=-\infty}^{\infty} \delta(\omega - k\omega_s)$$

- with 
$$\omega_s \triangleq \frac{2\tau}{T_s}$$

– Finally:

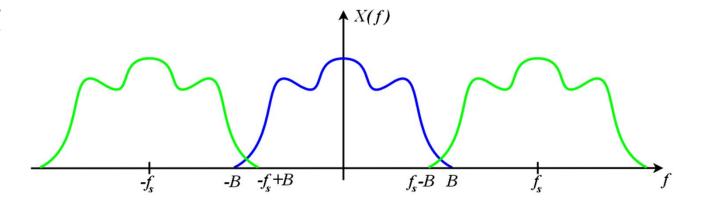
$$x(t) p(t) \stackrel{\mathcal{F}T}{\iff} \frac{1}{T_s} \sum_{k=-\infty}^{\infty} X \left( j(\omega - k\omega_s) \right)$$

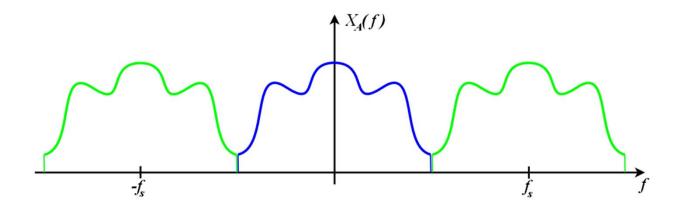
 By sampling, we'll have copies of the spectrum at harmonics of the sampling frequency



- Observations from the diagram
  - Choosing a large sampling frequency
    - Gaps between the copies of the spectrum
  - Choosing a small sampling frequency
    - Overlapping of the copies of the spectrum
    - Spectra overlapping: aliasing
    - Important concept

#### Aliasing





- Reconstruction (sampling)
  - Clearly assume no aliasing via proper selection of the sampling frequency: above Nyquest rate
  - We employ an ideal LPF to select the spectrum of the CT signal
- Note: quantization is not reversible!!
  - Quantization noise

#### Reconstruction

