

# Signal Sampling summary

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#### Contents



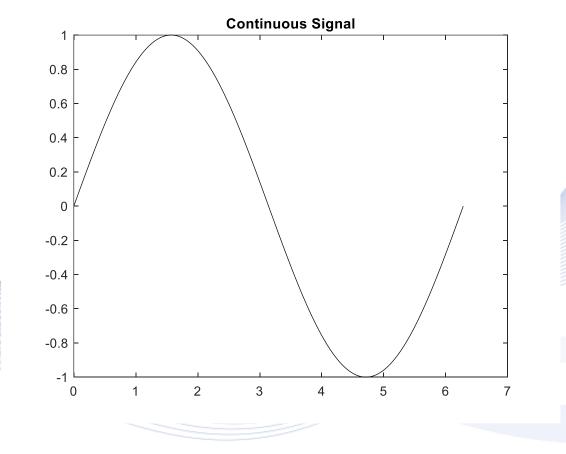
- Discrete/Continuous Signals
- Signal Sampling
- Signal Reconstruction
- Signal Quantization





#### **Continuous Signals**

Let the values of a signal x(t) be known for a continuous period of time, then we have a **continuous-time signal**  $x(t): \mathbb{R} \to \mathbb{R}.$ 



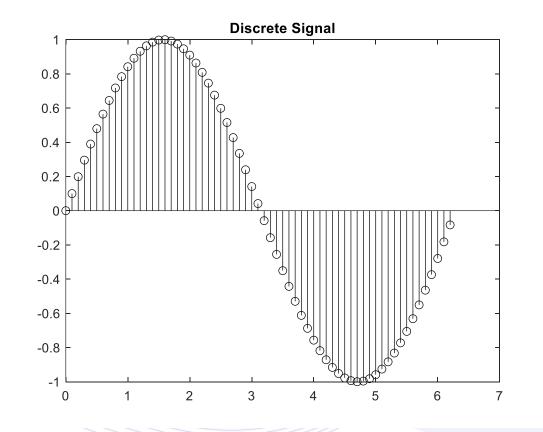


#### **Discrete Signals**



A *discrete-time* (*discrete*) signal is a time series (function) x(n):  $\mathbb{Z} \to \mathbb{R}$ , whose values are known only for some certain, discrete-time moments nT:

 $x(n) = x(nT), n \in \mathbb{Z}.$ 





### **Signal Digitization**

Signal digitization has two steps:

- Signal sampling at sampling frequency F.
- Signal quantization at b bit/sample.

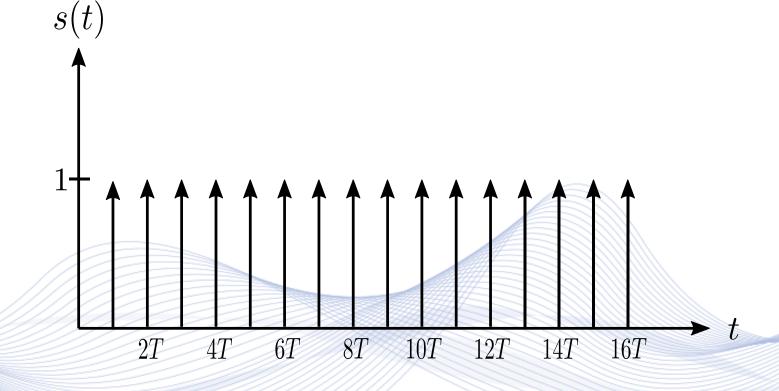
Signal digitization is performed by A/D converters.

$$x(t) \longrightarrow Sampling \qquad Quantization \qquad 
ightarrow x(n)$$





#### **Sampling Function**



Sampling function is a series of delta functions.



#### **Signal Sampling**



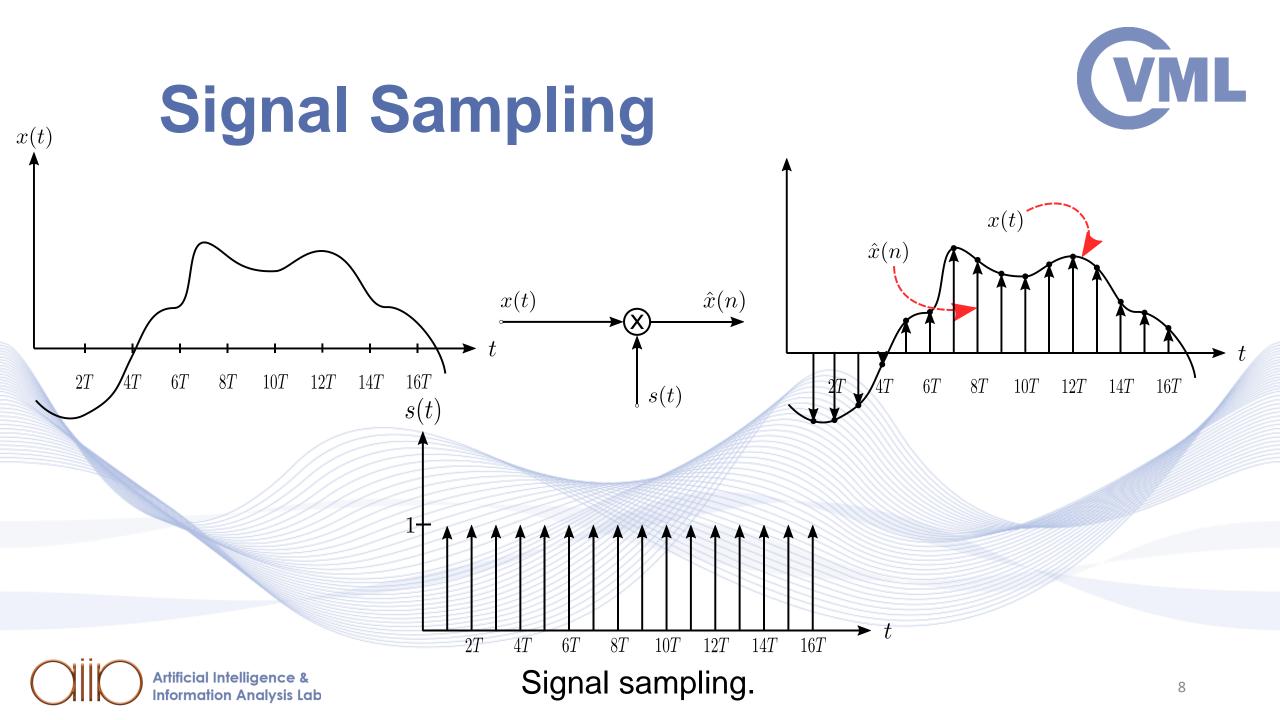
The output of a 1D **signal sampling** system is the discrete function  $\hat{x}(t)$ :

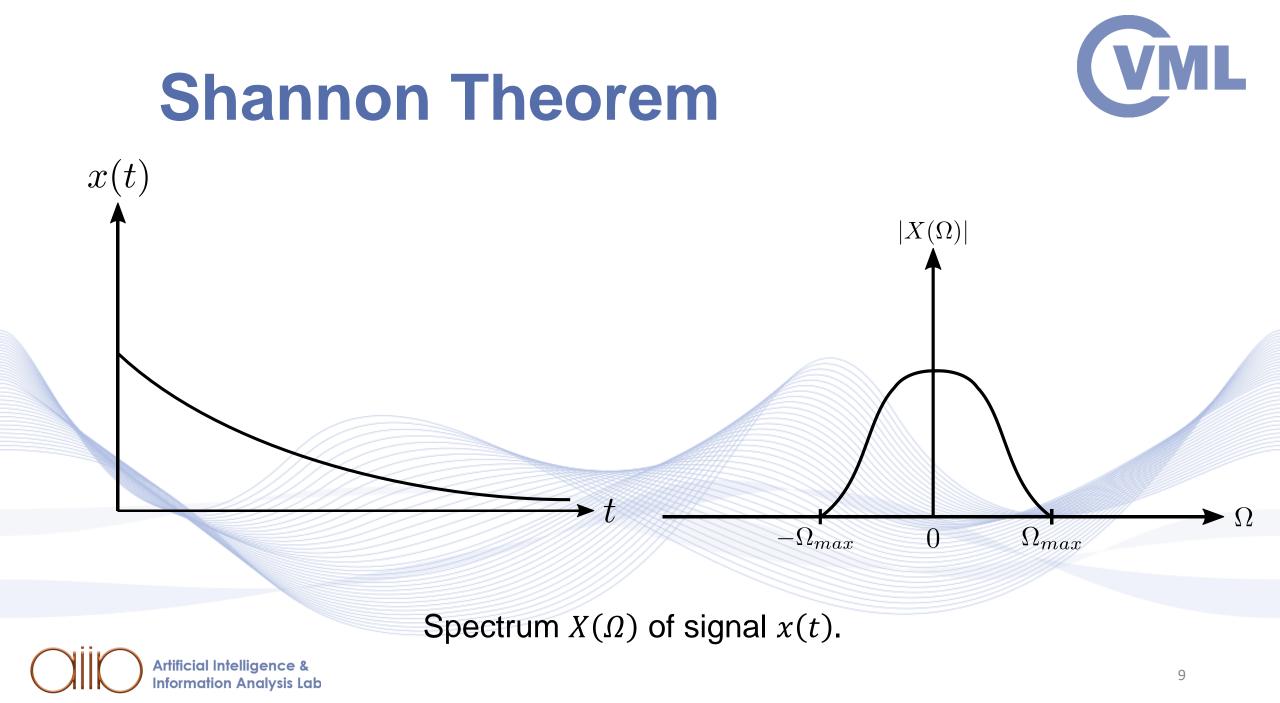
$$\hat{x}(t) = x(t)s(t) = \sum_{n=-\infty}^{\infty} x(t)\delta(t-nT) = \sum_{n=-\infty}^{\infty} x(nT)\delta(t-nT).$$

Equivalently, a discrete-time sampled signal is a function  $\mathbb{Z} \to \mathbb{R}$ :

 $\hat{x}(n) = x(nT), \qquad n \in \mathbb{Z}.$ 

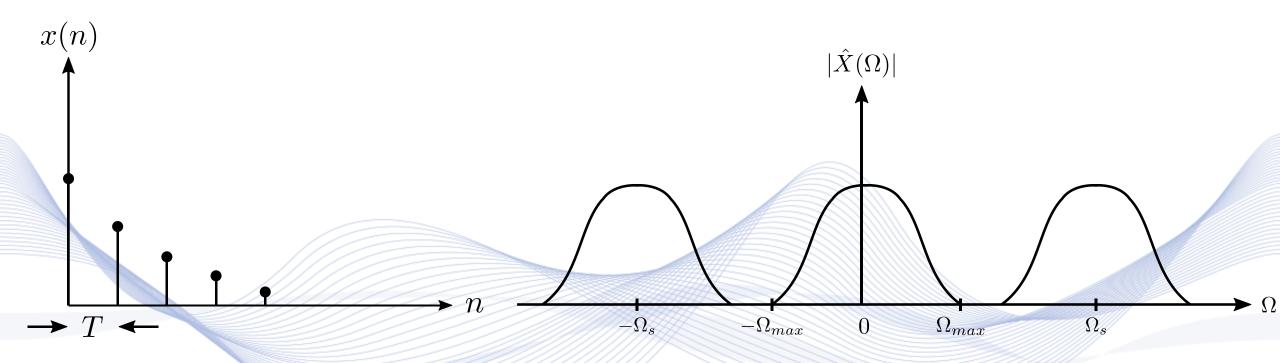








#### **Sampling without Aliasing**

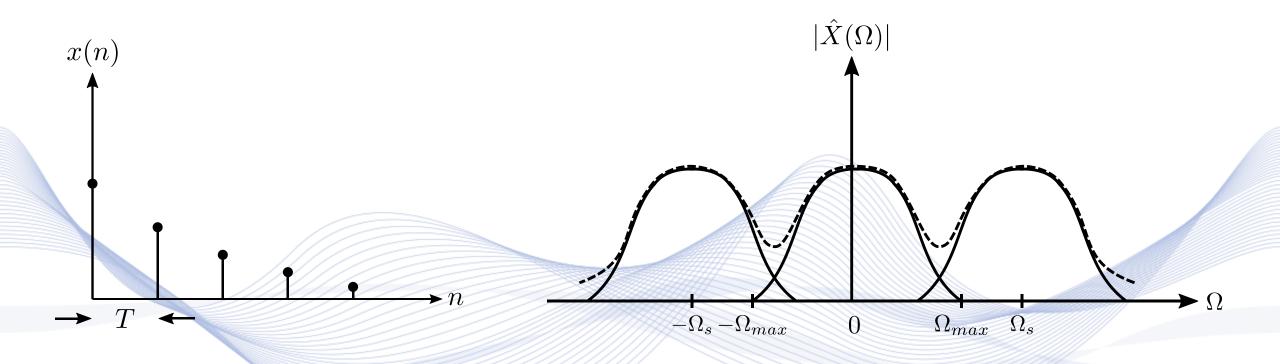


Sampled signal spectrum for  $\Omega_s \ge 2\Omega_{max}$ .

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#### **Sampling with Aliasing**



Sampled signal spectrum for  $\Omega_s < 2\Omega_{max}$ .

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#### **Shannon Theorem**



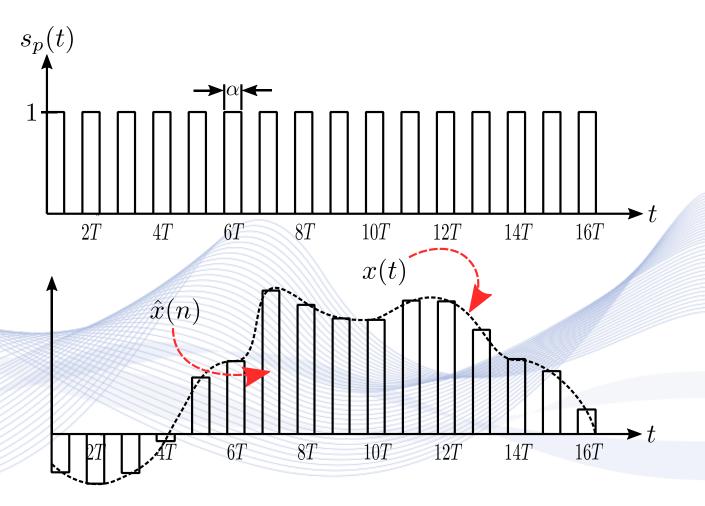
- The function x(t), of which the spectrum  $X(\Omega)$  has the highest frequency  $\Omega_{max}$ , is described absolutely with the values of x(nT),  $n = 0, \pm 1, \pm 2, ..., T = \frac{2\pi}{\Omega_s}$ , if and only if  $\Omega_s$  is greater than  $2\Omega_{max}$ .
- This special frequency  $(2\Omega_{max})$ , is called **Nyquist** frequency.





#### **Applied Signal Sampling**

It is practically impossible to implement ideal delta functions. Instead, sampling can be done using impulses of width  $\alpha$ .





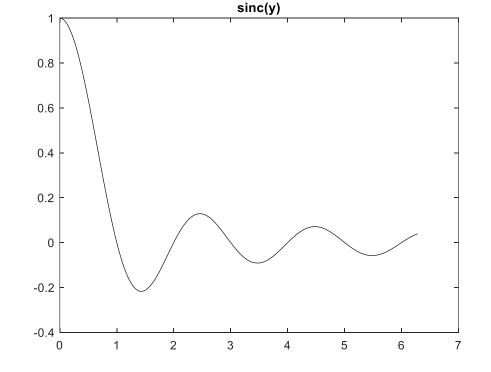


#### **Applied Signal Sampling**

• In that case, the spectrum of  $\widehat{X}(\Omega)$  is also periodically repeated, where coefficients  $C_n$  are given by:

$$C_n = \frac{1}{T} \int_{-\frac{\alpha}{2}}^{\frac{\alpha}{2}} e^{-in\Omega_s t} dt = \frac{\alpha}{T} \frac{\sin\left(\frac{n\Omega_s \alpha}{2}\right)}{\frac{n\Omega_s \alpha}{2}}.$$

• It is a sinc function:  $sinc(y) = \frac{sin(y)}{y}$  for:  $y = \frac{n\Omega_s \alpha}{2}$ .



 $-\Omega_{max}$ 

 $-\Omega_{s}$ 



If the sampling interval *T* is small enough, so that the spectrum  $X(\Omega)$  satisfies the relation:

π

 $\Omega_s$ 

 $|H(\Omega)|$ 

 $\Omega_{max}$ 

$$X(\Omega) = 0,$$
  $|\Omega| \ge \frac{\pi}{T},$ 

a square low-pass filter  $H(\Omega)$  can be used to recover  $X(\Omega)$  from the discrete 2D image spectrum  $\hat{X}(\Omega)$  by retaining only its basic period.

0



h(t)

 $\hat{\overline{T}}$ 

T

 The simplest *reconstruction filter* is the *sample-and-hold filter*, its impulse response is the following:

$$h(t) = \begin{cases} \frac{1}{T}, & 0 \le t \le T\\ 0, & t < 0, t > T. \end{cases}$$

Its use corresponds to 0-order polynomial interpolation.





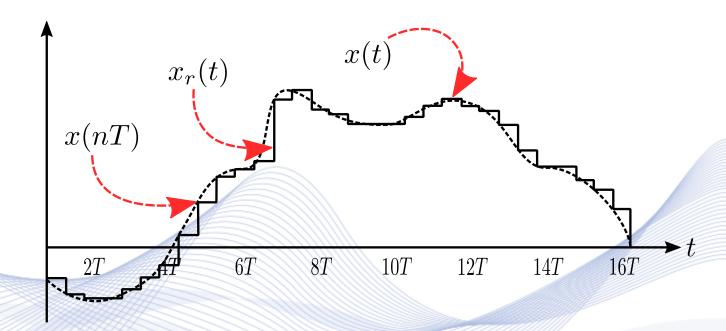
 $x(n) \bullet f = x_r(t)$ 

Elementary sample-and-hold circuit.





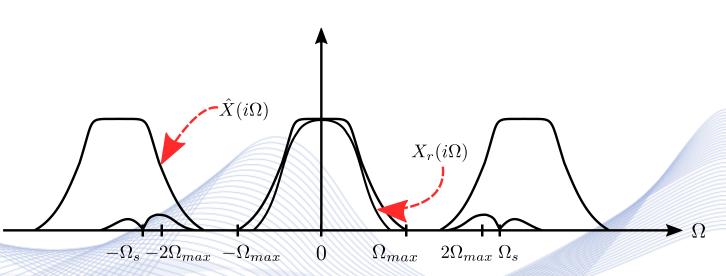
- The reconstructed signal  $x_r(t)$  has high frequencies due to discontinuities.
- It converges to the original signal x(t), as sampling period T tends to zero.







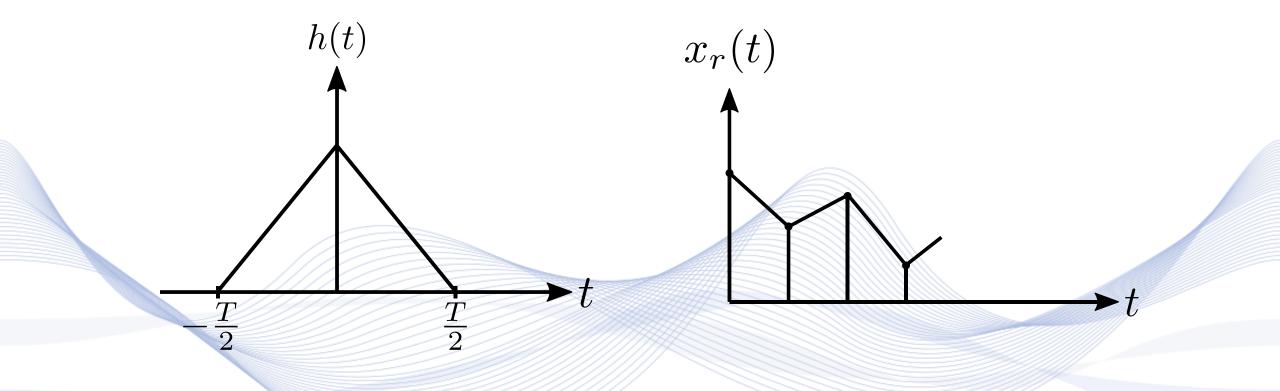
- The spectrum  $X_r(i\Omega)$  of the reconstructed signal contains residual high frequencies.
- Therefore, signal  $x_r(t)$  is usually filtered with a low-pass filter in order, to cut off the high frequency residues of  $X_r(i\Omega)$ .



Spectrum  $X_r(i\Omega)$  of the reconstructed signal.







Linear interpolation kernel and linear interpolation.





h(t)·t  $\frac{T}{2}$ 

Spline interpolation kernel and interpolation.



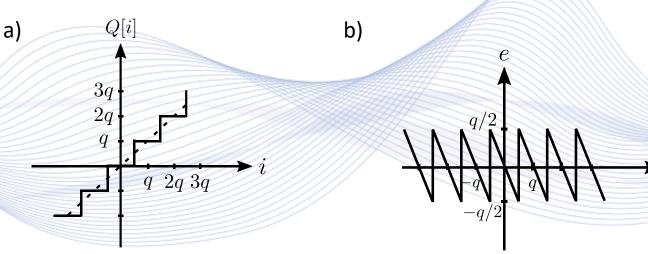


#### **Signal Quantization**

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*Quantization* is performed by an *A*/*D* converter.

- When b bits are allocated by signal sample, quantization step q is given by  $q = \frac{1}{2^b}$ ;
- Quantized signal levels:  $kq, k = 0, 1, 2, ..., 2^{b} 1$ .



a) Input-output curve of quantizer; b) Quantization error. 22 nformation Analysis Lab

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