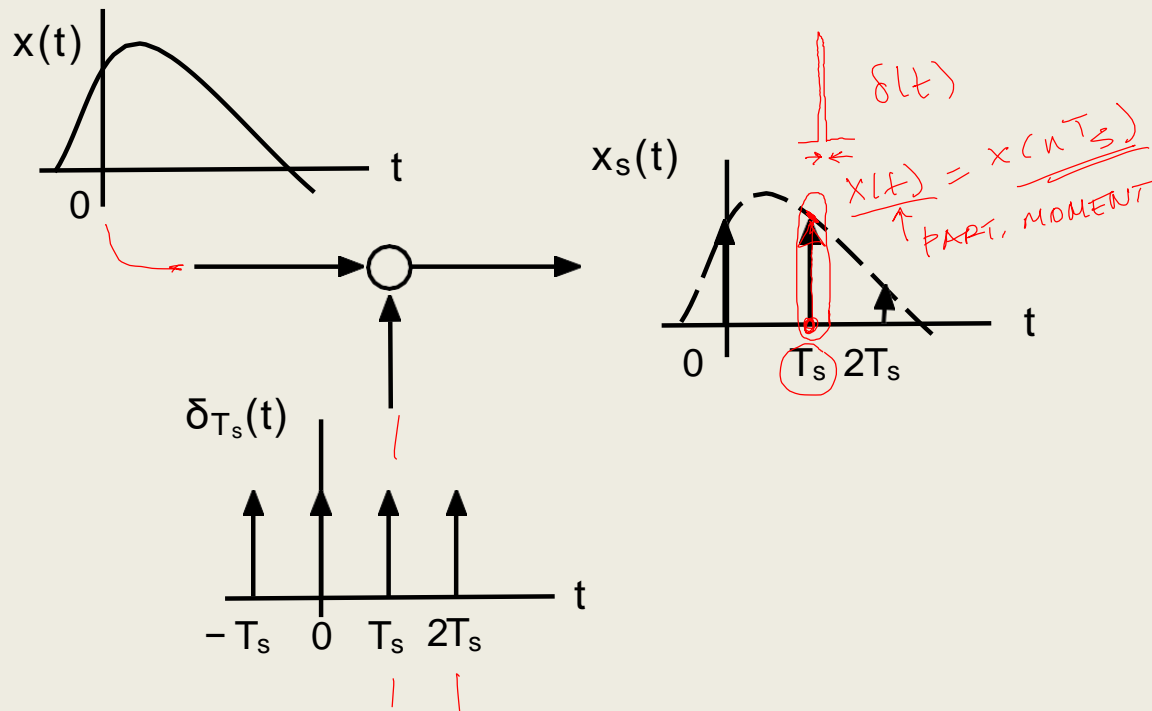


SIGNALS AND SYSTEMS USING MATLAB

Chapter 8 — Sampling Theory

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Uniform sampling



$$\delta_{T_s}(t) = \sum_n \delta(t - nT_s)$$

$$x_s(t) = x(t)\delta_{T_s}(t) = \frac{1}{T_s} \sum_{k=-\infty}^{\infty} x(t) e^{jk\Omega_s t}$$

- Modulation

$$\delta_{T_s}(t), \text{ periodic, } \Omega_s = 2\pi/T_s, \quad \delta_{T_s}(t) = \sum_{k=-\infty}^{\infty} D_k e^{jk\Omega_s t}$$

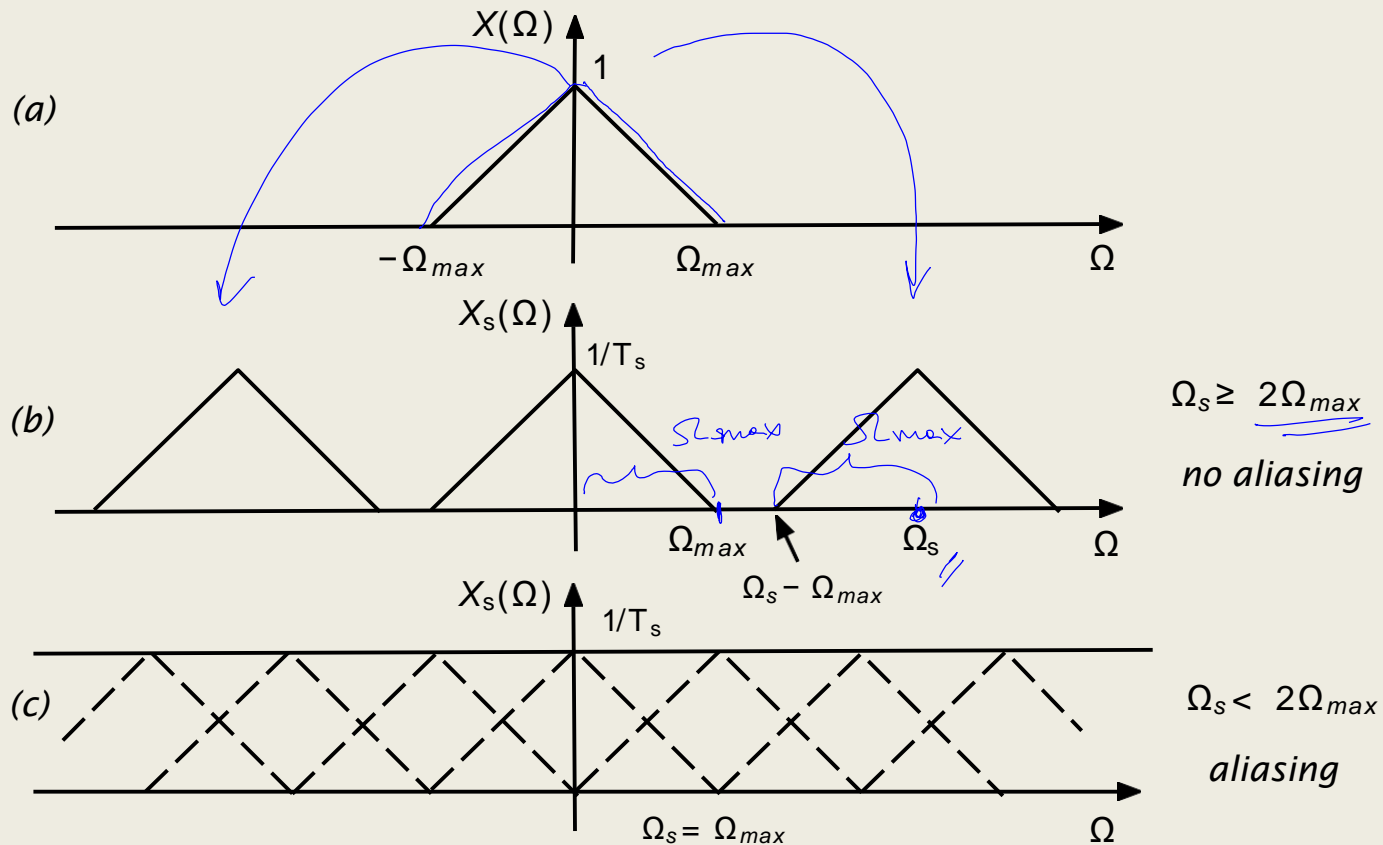
$$D_k = \frac{1}{T_s}, \quad x_s(t) = x(t)\delta_{T_s}(t) = \frac{1}{T_s} \sum_{k=-\infty}^{\infty} x(t) e^{jk\Omega_s t}$$

$$X_s(\Omega) = \frac{1}{T_s} \sum_{k=-\infty}^{\infty} \underline{X(\Omega - k\Omega_s)}$$

- Discrete-time Fourier transform

$$x_s(t) = \sum_{n=-\infty}^{\infty} \underline{x(nT_s)} \underline{\delta(t - nT_s)}$$

$$\underline{X_s(\Omega)} = \sum_{n=-\infty}^{\infty} \underline{x(nT_s)} \underline{e^{-j\Omega T_s n}}$$



(a) Spectrum of band-limited signal, (b) spectrum of sampled signal when satisfying the Nyquist sampling rate condition, (c) spectrum of sampled signal with aliasing (superposition of spectra, shown in dashed lines, gives a constant shown by continuous line)

Band-limited signals and Nyquist condition

A signal $x(t)$ is **band-limited** if its low-pass spectrum $X(\Omega)$ is such that

$$|X(\Omega)| = 0 \text{ for } |\Omega| > \Omega_{max}, \quad \Omega_{max} : \text{max frequency in } x(t)$$

can be **sampled uniformly and without frequency aliasing** using a sampling frequency

$$\Omega_s = \frac{2\pi}{T_s} \geq 2\Omega_{max} \quad \text{Nyquist sampling rate condition}$$

Example: $x(t) = 2 \cos(2\pi t + \pi/4), -\infty < t < \infty$, band-limited

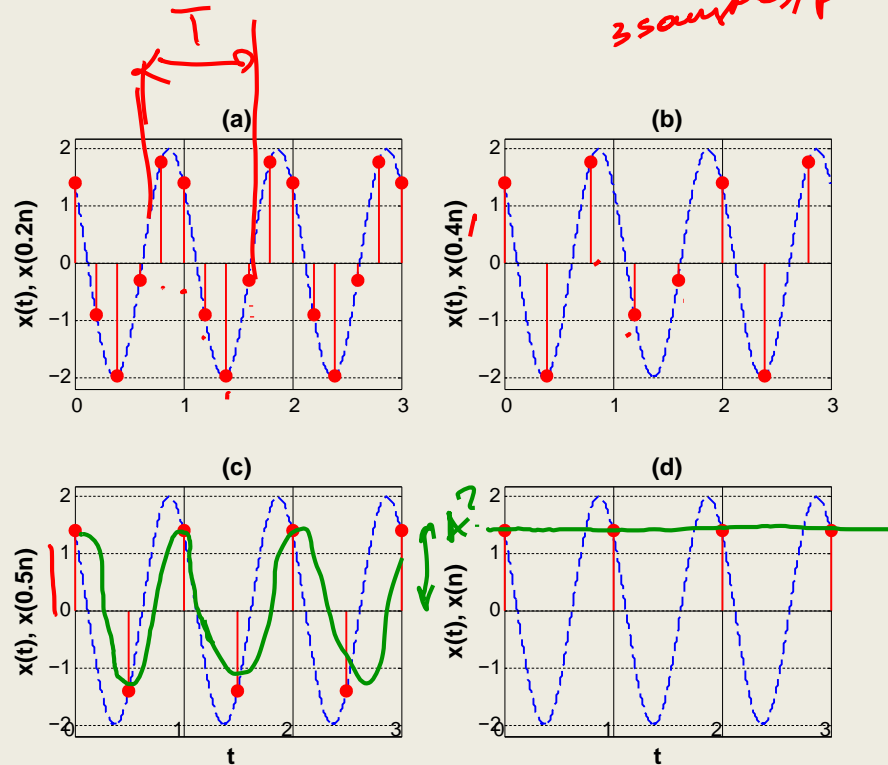
$$T_s = 0.4, \quad \Omega_s = 2\pi/T_s = 5\pi > 2\Omega_{max} = 4\pi, \text{ satisfy Nyquist}$$

$$\underline{x(nT_s)} = 2 \cos(2\pi \underbrace{0.4n}_{\substack{\text{wavy} \\ T_s \\ \text{wavy} \\ t}} + \pi/4) = 2 \cos\left(\frac{4\pi}{5} \underbrace{n}_{\substack{\text{circled} \\ \downarrow \text{SAMPLE} \\ \text{COUNTER}}} + \frac{\pi}{4}\right) \quad -\infty < n < \infty$$

$$T_s = 1, \quad \Omega_s = 2\pi < 2\Omega_{max} = 4\pi \quad \text{aliasing}$$

$$x(nT_s) = 2 \cos(2\pi n + \pi/4) = 2 \cos(\pi/4) = \sqrt{2}.$$

3 samples/period



Sampling of $x(t) = 2 \cos(2\pi t + \pi/4)$: (a) $T_s = 0.2$, (b) $T_s = 0.4$, (c) $T_s = 0.5$ and (d) $T_s = 1$ sec/sample

Nyquist–Shannon sampling theorem

Low-pass signal $x(t)$ is **band-limited** (i.e., $X(\Omega) = 0$ for $|\Omega| > \Omega_{max}$)

- Information in $x(t)$ preserved by sampled signal $x_s(t)$, with samples $x(nT_s) = x(t)|_{t=nT_s}$, $n = 0, \pm 1, \pm 2, \dots$, provided

sampling frequency $\Omega_s \geq 2\Omega_{max}$ (Nyquist sampling rate condition), or
sampling rate f_s (samples/sec) or sampling period T_s (sec/sample) are

$$f_s = \frac{1}{T_s} \geq \frac{\Omega_{max}}{\pi}$$

- When Nyquist condition is satisfied, $x(t)$ can be reconstructed by ideal low-pass filtering $x_s(t)$:

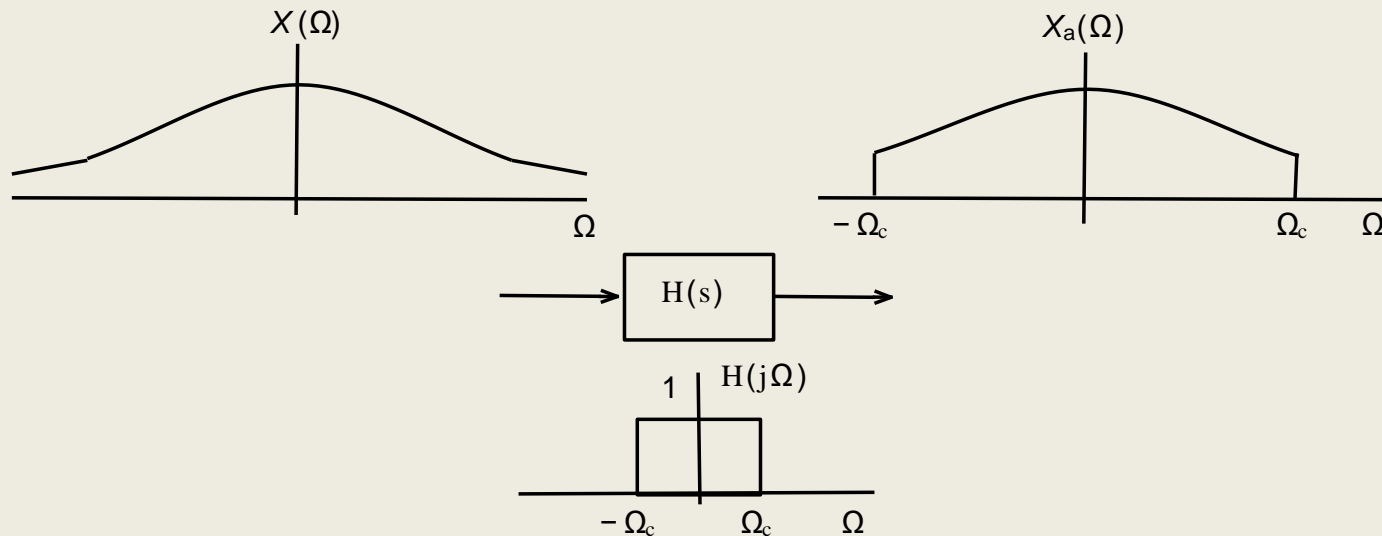
frequency response ideal LPF $H(j\Omega) = \begin{cases} T_s & -\Omega_s/2 < \Omega < \Omega_s/2 \\ 0 & \text{otherwise} \end{cases}$

Reconstructed (sinc interpolation)

$$x_r(t) = \sum_{n=-\infty}^{\infty} \underbrace{x(nT_s)}_{\text{sample}} \underbrace{\frac{\sin(\pi(t - nT_s)/T_s)}{\pi(t - nT_s)/T_s}}_{\text{sinc}}$$

Antialiasing filtering

For signals that do not satisfy the band-limitedness condition



Anti-aliasing filtering of non band-limited signal

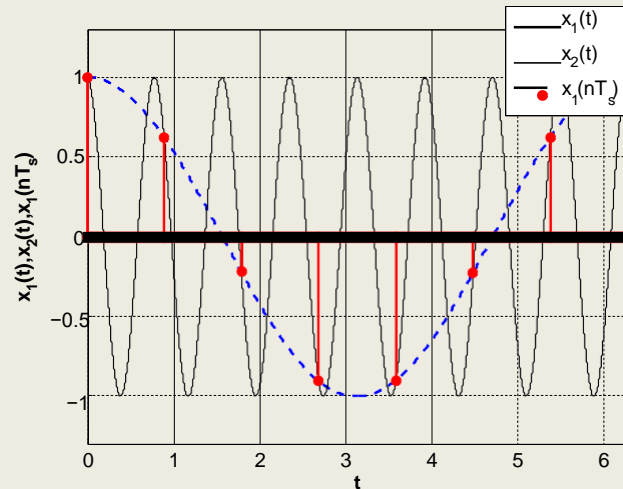
Example: Aliasing effects

$$x_1(t) = \cos(\Omega_0 t), \quad x_2(t) = \cos((\Omega_0 + \Omega_1)t) \quad \Omega_1 > 2\Omega_0$$

sampling signals with $T_s = 2\pi/\Omega_1$

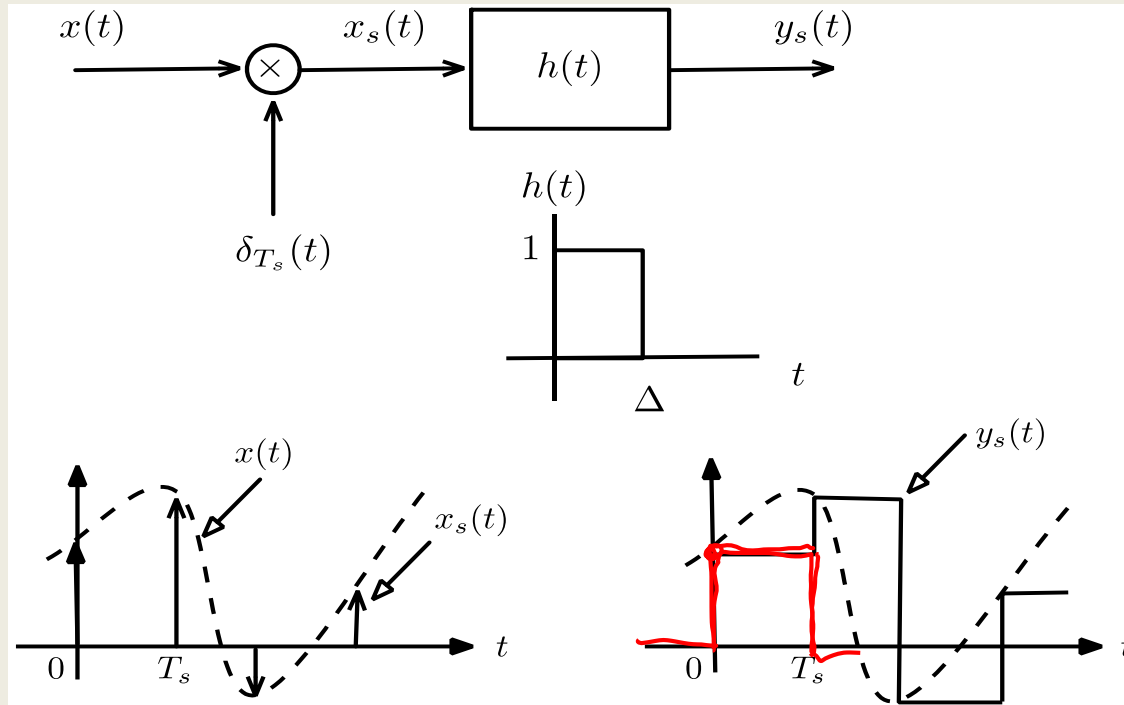
$$x_1(nT_s) = \cos(\Omega_0 nT_s), \quad x_2(nT_s) = \cos((\Omega_0 + \Omega_1)nT_s) = \cos(\Omega_0 T_s n) = x_1(nT_s)$$

No frequency aliasing in $x_1(nT_s)$, frequency aliasing in $x_2(nT_s)$



Sampling sinusoids of frequencies $\Omega_0 = 1$ and $\Omega_0 + \Omega_1 = 8$ with $T_s = 2\pi/\Omega_1$. The higher frequency signal is under-sampled, causing aliasing and making the two sampled signals coincide

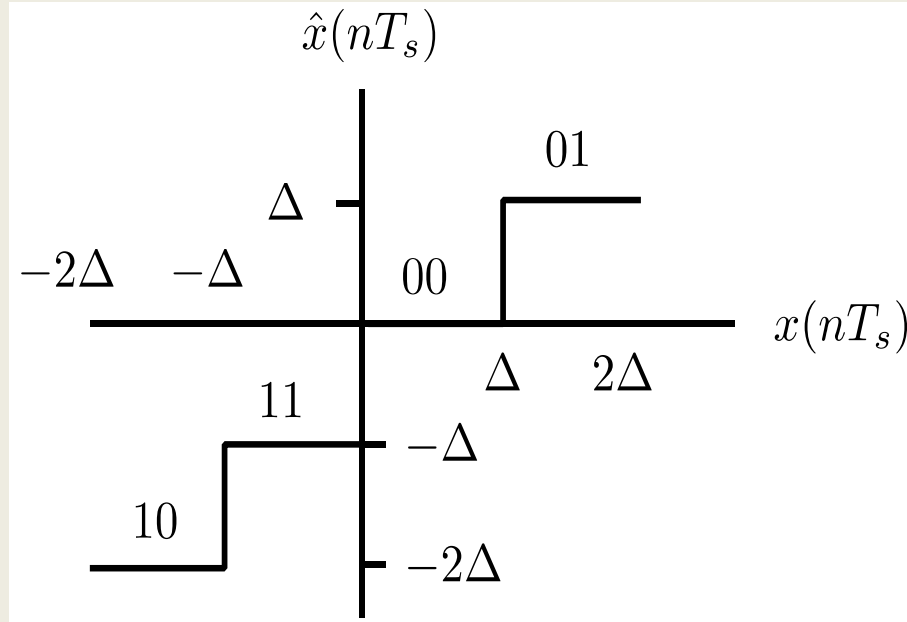
Practical aspects of sampling — Sample-and-hold sampling



Sample-and-hold sampling system for $\Delta = T_s$; $y_s(t)$ multi-level signal

$$y_s(t) = (x_s * h)(t) \Rightarrow Y_s(\Omega) = \left[\frac{1}{T_s} \sum_k X(\Omega - k\Omega_s) \right] \frac{\sin(\Delta\Omega/2)}{\Omega/2} e^{-j\Omega\Delta/2}$$

Practical aspects of sampling — Quantization and coding



n-bit

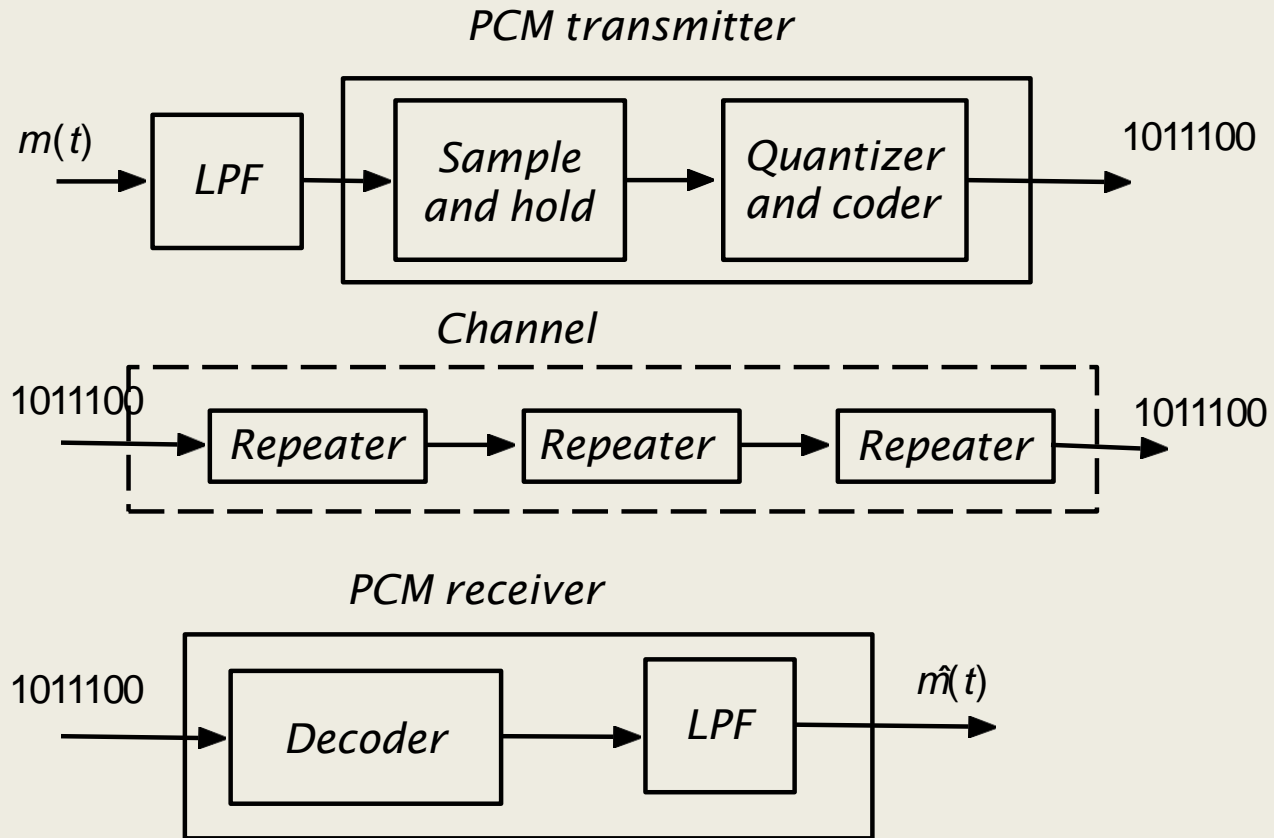
Four-level quantizer and coder.

Sampled signal $x(nT_s) = x(t)|_{t=nT_s}$

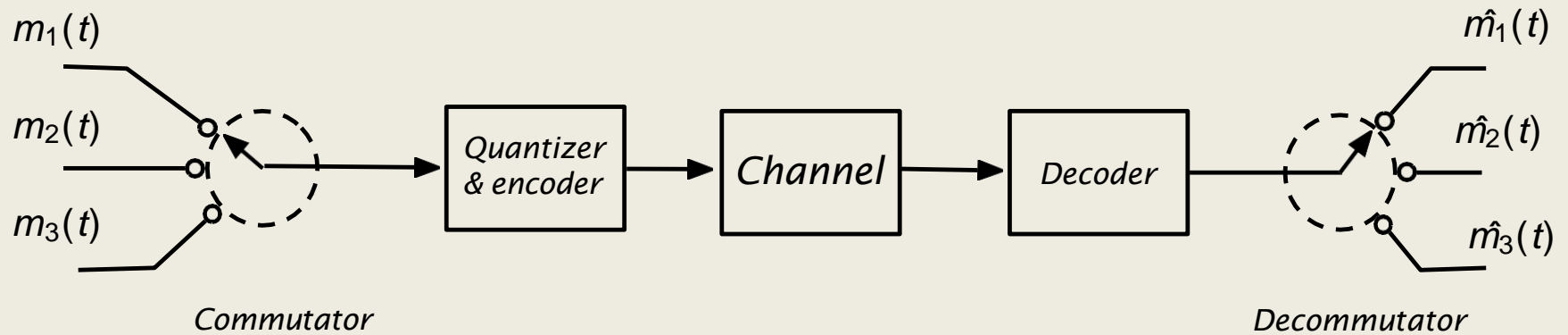
four-level quantizer: $k\Delta \leq x(nT_s) < (k+1)\Delta \Rightarrow \hat{x}(nT_s) = k\Delta, \quad k = -2, -1, 0, 1$

coder assigns binary number to each output level of quantizer

Application to digital communications



PCM system: transmitter, channel and receiver.



Time Division Multiplexing (TDM) system: transmitter, channel and receiver