

Introduction to Digital Modulation



EE4367 Telecom. Switching & Transmission

Prof. Murat Torlak

Introduction

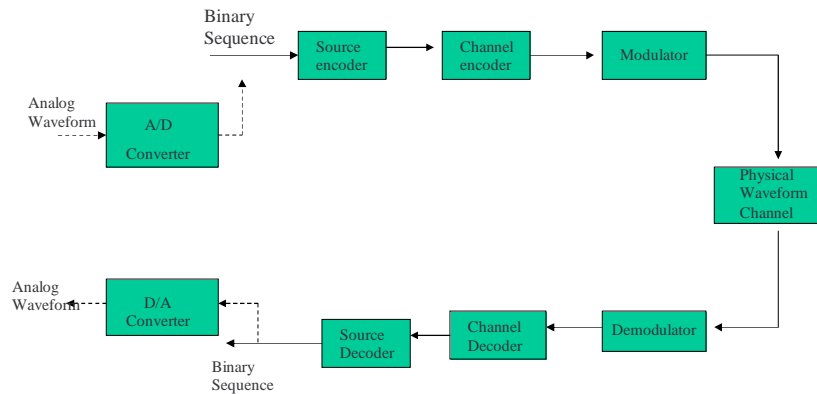
- ❑ In a digital communication system, the source to be transmitted is discrete both in time and amplitude
- ❑ Digital information carrying signals must be first converted to an analog waveform prior to transmission
- ❑ At the receiving end, analog signals are converted back to a digital format before presentation to the end user
- ❑ The conversion process at the transmitting end is known as modulation
- ❑ The receiving end is known as demodulation or detection



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Functional Block Diagram of a Binary Digital Communication System



Digital Modulation

□ Overview:

- In digital wireless communication systems, the modulating signal may be represented as a time sequence of symbols or pulses, where each symbol has m finite states. Each symbol represents n bits of information where $n = \log_2 m$ bits/symbol.

□ Advantages of Digital over Analog:

- Greater noise immunity (due to its finite process)
- Robustness to channel impairments
- Easier multiplexing of various forms of information like voice, data, video



Digital Modulation

- ❑ Security - by using coding techniques to avoid jamming
- ❑ Accommodation of digital error control codes which detect and/or correct transmission errors
- ❑ Equalization to improve the performance of over all communication link
- ❑ Supports complex signal conditioning and processing methods



Digital Modulation

- ❑ Factors that influence Digital Modulation:
 - ❑ Low BER at low received SIR
 - ❑ performs well in multi-path and fading
 - ❑ High spectral efficiency
 - ❑ The performance of a modulation scheme is often measured in terms of its power efficiency and bandwidth efficiency
 - ❑ The power efficiency is the ability of a modulation technique to preserve the fidelity (acceptable BER) of the digital message at low power levels
 - ❑ (Good BER performance at a low SIR under conditions of co-channel interference, fading, and time dispersion)



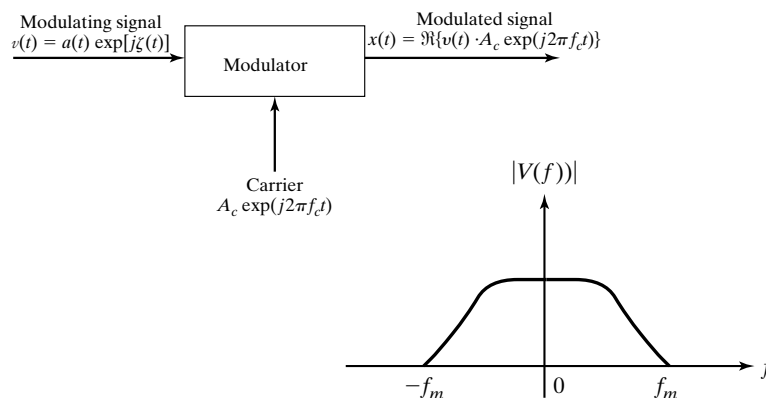
Digital Modulation

- ❑ The source information is normally represented as a baseband (low-pass) signal
- ❑ Because of signal attenuation, it is necessary to move the baseband signal spectrum to reside at a much higher frequency band centered at f_c , called the carrier frequency, in the radio spectrum
- ❑ At the receiver end, the demodulation process removes the carrier frequency to recover the baseband information signal
- ❑ Choose different carrier frequencies for different signals
 - ❑ Modulation/demodulation process facilitates channel assignment and reduces interference from other transmissions.



Generic Bandpass Transmission

- ❑ A_c is a constant denoting the amplitude carrier



Modulation Categories

- The modulated signal, $x(t)$, is given by

$$\begin{aligned}x(t) &= \Re\{v(t)A_c \exp(j2\pi f_c t)\} \\ &= A_c a(t) \cos(2\pi f_c t + \zeta(t)) \\ &= A_c a(t) \cos \zeta(t) \cos(2\pi f_c t) - A_c a(t) \sin \zeta(t) \sin(2\pi f_c t).\end{aligned}$$

- The modulation can be classified into two categories:
 - Linear modulation: A modulation process is *linear* when both $a(t)\cos\zeta(t)$ and $a(t)\sin\zeta(t)$ terms are linearly related to the message information signal.
 - Nonlinear modulation: when the modulating signal, $v(t)$, affects the frequency of the modulated signal. The definition of nonlinear is that superposition does not apply.



Modulation Examples

- Examples of linear modulation include
 - amplitude modulation, where the modulating signal affects only the amplitude of the modulated signal (i.e., when $\zeta(t)$ is a constant $\forall t$ (for any t)), and
 - phase modulation (with a rectangular phase shaping function) where the modulating signal affects only the phase of the modulated signal (i.e., when $\zeta(t)$ is a constant over each signaling (symbol) interval and $a(t)$ is a constant $\forall t$).
- Example of the nonlinearly modulated signal is

$$x(t) = a \cdot A_c \cos(2\pi f_c t + \zeta(t))$$

where the angle, $\zeta(t)$ is the integral of a frequency function



Binary Digital Modulation Examples

Binary amplitude shift keying (ASK): $\zeta(t)=0$.

- amplitude component $a(t)$:
- $a(t)=1$ for symbol "1" and $a(t)=0$ for symbol "0".
- The modulated signal is

$$x(t) = \begin{cases} A_c \cos(2\pi f_c t), & \text{symbol "1"} \\ 0, & \text{symbol "0"} \end{cases}$$

Binary phase shift keying (PSK): $a(t)=1$

- the phase component $\zeta(t)$:
- $\zeta(t)=0$ for symbol "1" and
- $\zeta(t)=\pi$ for symbol "0"
- The modulated signal is

$$x(t) = \begin{cases} A_c \cos(2\pi f_c t), & \text{symbol "1"} \\ A_c \cos(2\pi f_c t + \pi) = -A_c \cos(2\pi f_c t), & \text{symbol "0"} \end{cases}$$



Binary Digital Modulation Examples

Binary frequency shift keying (FSK): $a(t)=1$

- $\zeta(t)$: $\zeta(t)=2\pi(\Delta/2)t + \phi_1$ for symbol "1"
- $\zeta(t)=-2\pi(\Delta/2)t + \phi_2$ for symbol "0"
- where Δ is the frequency separation between the signals for symbols "1" and "0" respectively, ϕ_1 and ϕ_2 are any constants in $[-\pi, +\pi]$.
- The modulated signal is

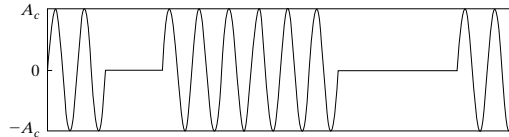
$$x(t) = \begin{cases} A_c \cos[2\pi(f_c + 0.5\Delta)t + \phi_1], & \text{symbol "1"} \\ A_c \cos[2\pi(f_c - 0.5\Delta)t + \phi_2], & \text{symbol "0"} \end{cases}$$

- The instantaneous frequency of the modulated signal $x(t)$ is
 - $f(t)=f_c+0.5\Delta$ for symbol "1" and $f(t)=f_c-0.5\Delta$ for symbol "0".
- The instantaneous phase of $x(t)$ is given by

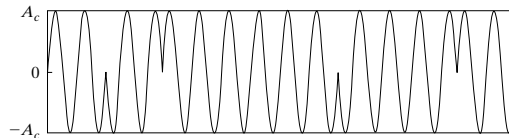
$$\phi(t) = \int_{-\infty}^t 2\pi f(\tau) d\tau.$$



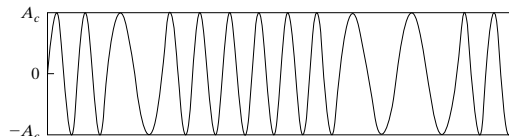
Modulated Signal Waveforms



(a) amplitude-shift keying



(b) phase-shift keying

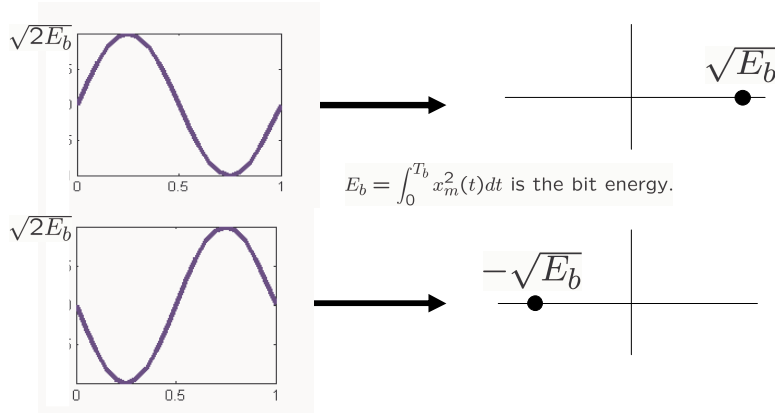


(c) frequency-shift keying



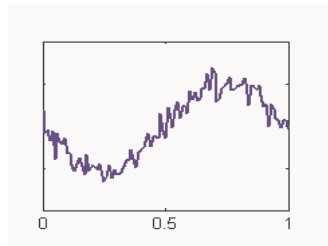
Constellation Representation

□ Typical signal waveforms for BPSK transmission and constellation

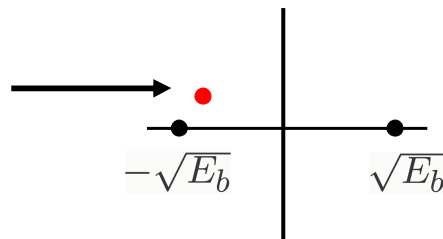


Decision Regions

- Waveforms in noise
- Noise adds uncertainty to the location of the signal state



Boundary of two decision regions

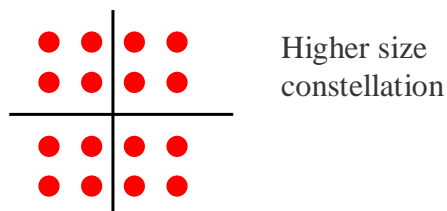


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Signal Space and Decision Regions

- Time domain \rightarrow Signal space domain representation
- Vector-Space Representation of M -ary Signals
- The digital source generates digital symbols for transmission at a rate of R_s symbols per second.
- The symbols are taken from an alphabet of size $M=2^l$
- Each symbol can be represented by l binary digits.
- The transmission rate: $R_b=lR_s$ bits per second (bps), where R_b is the bit rate.



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Common Digital Modulation Techniques

- ❑ M-ary Phase Shift Keying (MPSK):
- ❑ During the signaling interval, T_s , one of the waveforms is selected

$$x_m(t) = \sqrt{\frac{2E_s}{T_s}} \cos\left(2\pi f_c t + \frac{2\pi(m-1)}{M}\right), \quad 0 \leq t \leq T_s, \quad m = 1, 2, \dots, M,$$

where E_s is the symbol energy given by

$$E_s = \int_{-\infty}^{\infty} x_m^2(t) dt$$

- ❑ As each waveform represents l binary digits, we have $E_s = lE_b$ where E_b is the bit energy.



Differential PSK (DPSK)

- ❑ DPSK is noncoherent form of phase shift keying which avoids the need for a coherent reference signal at the receiver.
- ❑ Input binary sequence is first differentially encoded and then modulation using a BPSK modulator.
- ❑ Differentially encoded sequence $\{d_k\}$ is generated from the input binary sequence $\{m_k\}$ by complementing the modulo-sum of m_k and d_{k-1}
 - ❑ Effect is to leave the symbol d_k unchanged from the previous symbol if m_k is 1, and toggle d_k if m_k is 0.

$$d_k = m_k \oplus d_{k-1}$$

- ❑ Decoding

$$m_k = \overline{d_k} \oplus d_{k-1}$$

- ❑ Demodulation

$$x_k = \overline{r_k} r_{k-1}$$

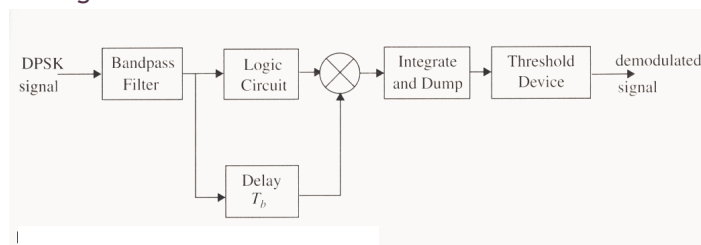


DPSK

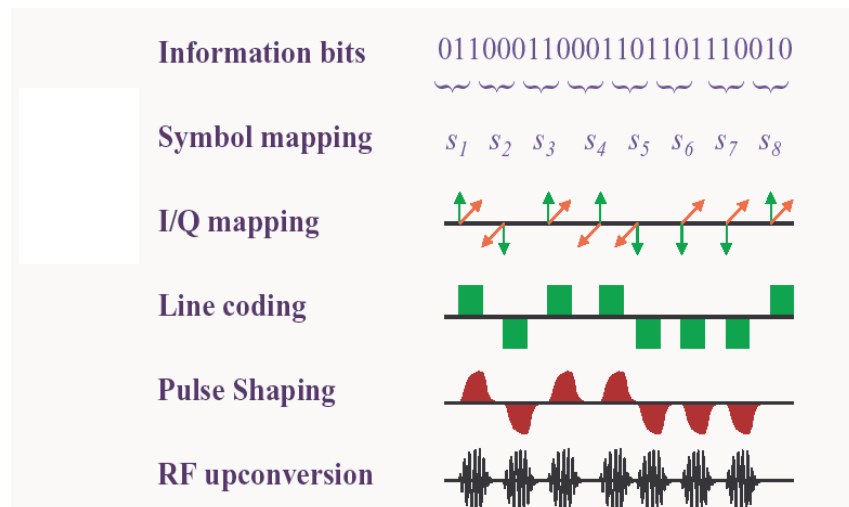
- Illustration of the differential encoding process

$\{m_k\}$		1	0	0	1	0	1	1	0
$\{d_{k-1}\}$		1	1	0	1	1	0	0	0
$\{d_k\}$	1	1	0	1	1	0	0	0	1

- Block diagram of DPSK receiver



Digital Modulation Stages



Pulse Shaping → Bandwidth Limitation

- Why pulse shaping?
 - ISI can be minimized by increasing the channel bandwidth
 - Mobile communication systems operate with minimal BW
 - Hence pulse shaping techniques are used to reduce ISI and spectral BW
- Nyquist criterion for ISI Cancellation:
 - Effects of ISI could be completely nullified if, at every sampling instant, the response due to all symbols except the current symbol is made equal to zero

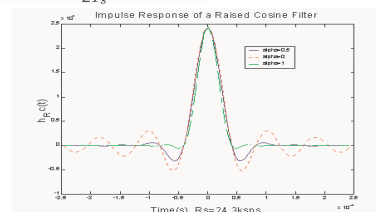
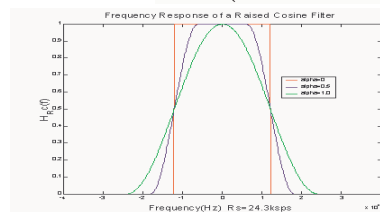
$$p(nT_s) = \begin{cases} 1 & n = 0 \\ 0 & n \neq 0 \end{cases}$$



Raised Cosine Filter

- Raised Cosine Roll-off filter:
 - It satisfies the Nyquist criterion
 - The spectral efficiency offered by raised cosine filter only occurs if exact pulse shape is preserved at the carrier
- The transfer function of raised cosine filter

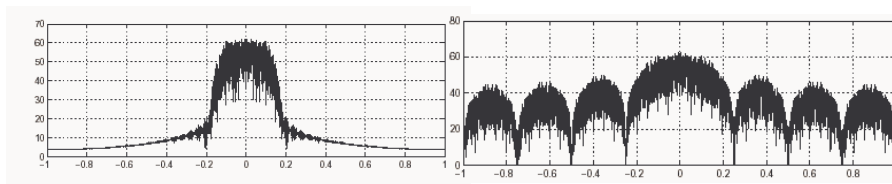
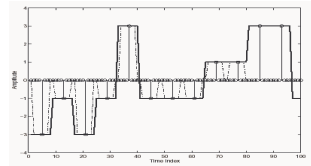
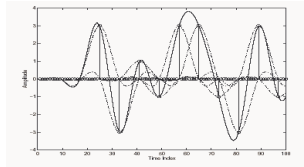
$$|H(f)| = \begin{cases} 1, & 0 \leq f \leq \frac{1-\alpha}{2T_s} \\ \frac{1}{2} \left\{ 1 - \sin \left[\frac{\pi(2fT_s - 1)}{2\alpha} \right] \right\}, & \frac{1-\alpha}{2T_s} \leq f \leq \frac{1+\alpha}{2T_s} \\ 0, & f > \frac{1+\alpha}{2T_s} \end{cases}$$



Signal Spectrum after Pulse Shaping

After pulse-shaping

before pulse-shaping



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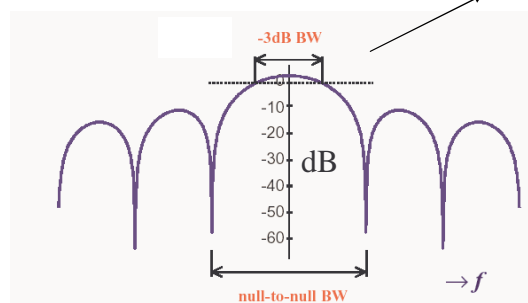
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Bandwidth Definitions

Measures of Bandwidth (BW):

- 99% BW → freq. range where 99% of power is
- Absolute BW : Range of frequencies over a non-zero spectrum
- Null-to-Null BW : Width of the main spectral lobe
- Half-power bandwidth: 3dB bandwidth

Half-power bandwidth

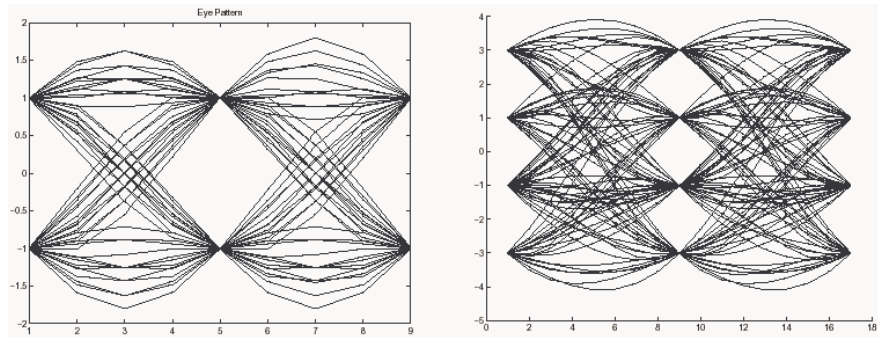


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Eye Diagram

- Below figures show an eye diagram:
 - Examples of 2-level and 4-level eye patterns
- The wider the “eye” opens, the better the signal quality is.
- Empirical measure of the quality of the received signal.

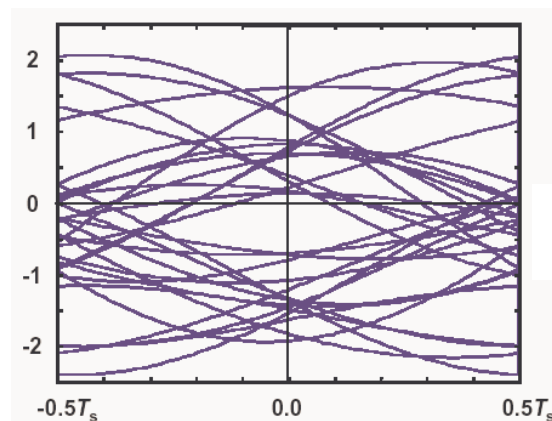


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Intersymbol Interference

- Intersymbol interference channel



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Probability of Transmission Error

- Coherent Reception in an AWGN Channel
- BPSK: The transmitted signal

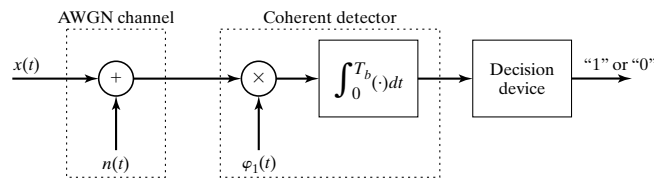
$$x(t) = \begin{cases} \sqrt{E_b}\varphi_1(t), & \text{for symbol "1"} \\ -\sqrt{E_b}\varphi_1(t), & \text{for symbol "0"} \end{cases}$$

- In an AWGN channel, the received signal is

$$r(t) = x(t) + n(t)$$

- where $n(t)$ represents the white Gaussian noise process with zero mean and two-sided psd $N_0/2$.

- Coherent reception of BPSK in an AWGN channel



Probability of Bit Error

- The conditional probability at the output of correlator:

$$f_{r_i}(x) = \frac{1}{\sqrt{2\pi}\sigma_N} \exp\left[-\frac{(x - \mu_i)^2}{2\sigma_N^2}\right], \quad i = 0, 1; -\infty < x < \infty.$$

- The decision rule is as follows:

$$\begin{cases} \text{if } r \geq 0, \text{ symbol "1" was sent;} \\ \text{if } r < 0, \text{ symbol "0" was sent.} \end{cases}$$

- With equally likely symbols "1" and "0", the probability of symbol (bit) error,

$$\begin{aligned} P_b &= P(r \geq 0 | \text{symbol "0" was sent})P(\text{symbol "0" was sent}) \\ &\quad + P(r < 0 | \text{symbol "1" was sent})P(\text{symbol "1" was sent}) \\ &= \frac{1}{2} \left\{ \int_0^{\infty} \frac{1}{\sqrt{2\pi}\sigma_N} \exp\left[-\frac{(x - \mu_0)^2}{2\sigma_N^2}\right] dx + \int_{-\infty}^0 \frac{1}{\sqrt{2\pi}\sigma_N} \exp\left[-\frac{(x - \mu_1)^2}{2\sigma_N^2}\right] dx \right\} \\ &= Q\left(\sqrt{\frac{2E_b}{N_0}}\right), \quad \text{where } Q(t) = \frac{1}{\sqrt{2\pi}} \int_t^{\infty} \exp(-z^2/2) dz \end{aligned}$$

