

EE3414

Multimedia Communication Systems - I

Digital Communications Principles

Based on Lecture Notes by Elza Erkip

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Outline

- Digital Communication Systems
 - Modulation of digital signals
 - Error probability vs. SNR
 - Error correction coding
 - Channel capacity: noiseless case, and noisy case
 - Advantages of digital communication

How to send digital signals?

- Digital bits -> analog waveforms (digital modulation)
 - Used in telephone modems, cell phones, digital TV, etc.
- Digital bits -> digital pulse sequences (line coding)
 - Used in computer networks
- How do we deal with channel noise?
 - Error detection (e.g. parity check)
 - Error correction coding
- How fast can we send bits ?
 - Channel capacity depends on bandwidth, modulation, and SNR
 - Shannon channel capacity formula

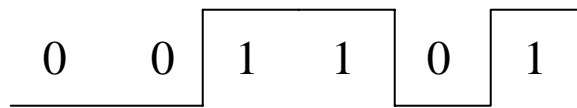
Modulation of Digital Signals

- For transmission of digital bits over analog channels
 - Convert group of digital bits into analog waveforms (symbols)
 - The analog waveforms are formed by adapting the amplitude and/or phase of a carrier signal (ASK or PSK)
 - The carrier frequency is chosen based on the desired/acceptable operating range of the channel
 - An analog channel of bandwidth B can carry at most $2B$ symbols/s. For reduced inter-symbol interference, lower than $2 \cdot B$ symbol rate is used typically
 - Shannon's capacity formula characterizes the dependency of channel capacity on channel bandwidth and noise level
 - Equalizer is used at the receiver to reduce the inter-symbol interference

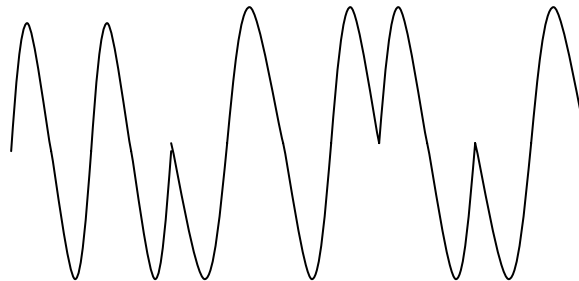
A Simple Example

- Digital information: Sequence of 0's and 1's: 001101.....
- One bit every T seconds. During $0 < t < T$
 - To send a 0, send $s_0(t) = A\cos(2\pi f_c t)$
 - To send a 1, send $s_1(t) = -A\cos(2\pi f_c t)$

- Input signal



- Modulated signal



- This is called **Binary Pulse Amplitude Modulation (PAM)** or **Binary Phase Shift Keying (BPSK)**.
- For a channel with bandwidth B, $T \geq 1/2B$, to avoid inter-symbol interference

Amplitude Shift Keying (ASK)

M-ary ASK: each group of $\log_2 M$ bits generates a symbol. The number corresponding to the symbol controls the amplitude of a sinusoid waveform. The number of cycles in the sinusoid waveform depends on the carrier frequency.
(Also known as Pulse Amplitude Modulation or PAM)

4-ASK: 2 bits/symbol (00=-3, 01=-1, 11=1, 10=3)



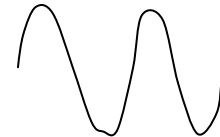
"00"(-3A)



"01"(-A)



"11" (A)



"10"(3A)

Example: Given a sequence: 01001011..., what is the analog form resulting from 4-ASK?

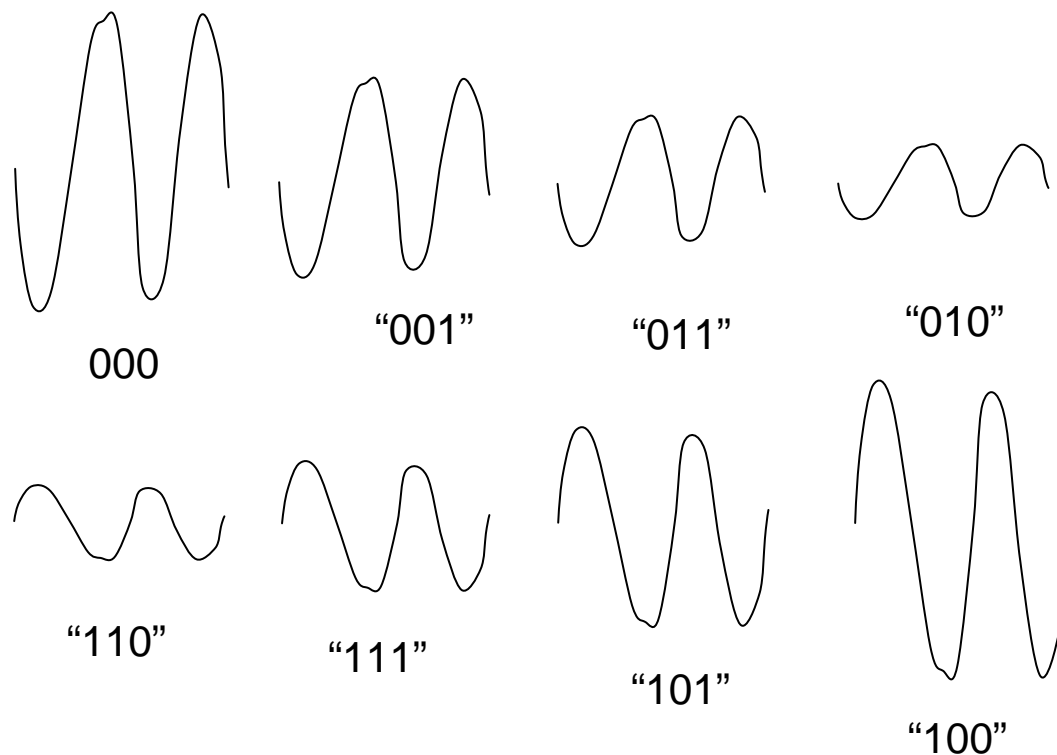
Symbol representation: "-1", "-3", "3", "1"

Waveform:



8-ASK

8-ASK: 3 bits/symbol (000=-7, 001=-5, 011=-3, 010=-1, 110=1, 111=3, 101=5, 100=7)

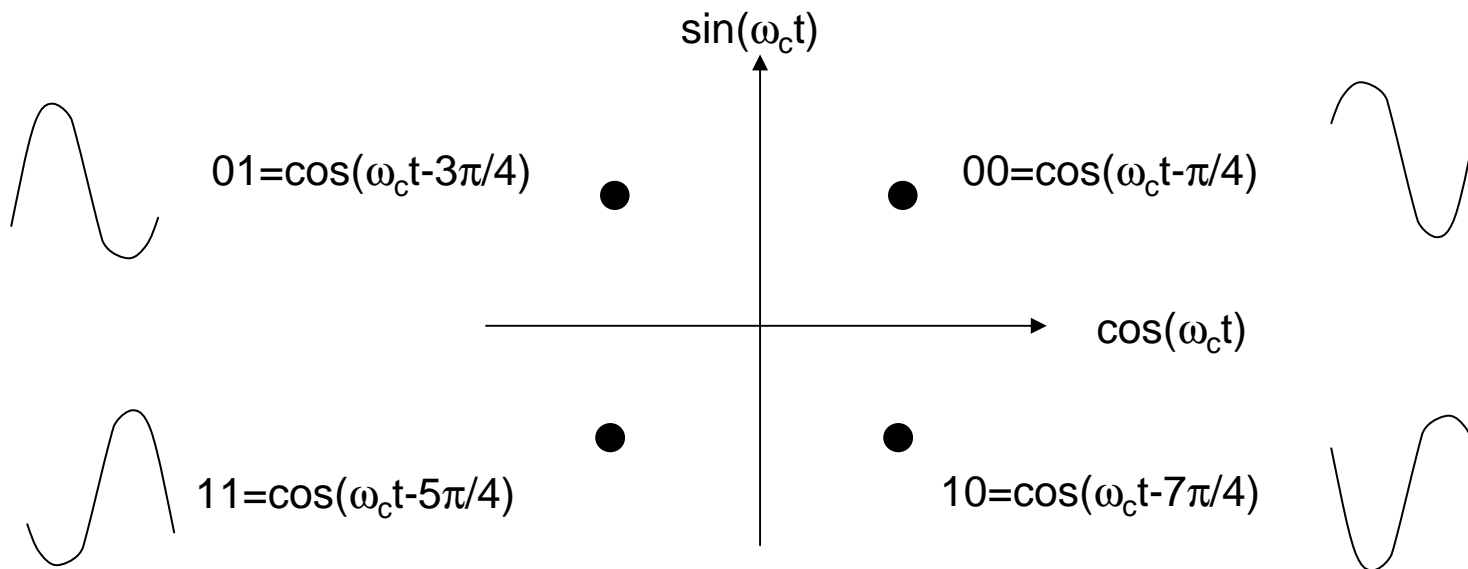


The mapping from bits to symbols are done so that adjacent symbols only vary by 1 bit, to minimize the impact of transmission error (this is called **Gray Coding**)

Quadrature Amplitude Modulation (QAM)

M-ary QAM uses symbols corresponding to sinusoids with different amplitude as well as phase, arranged in the two-dimensional plane.

Ex. 4-QAM (only phase change):

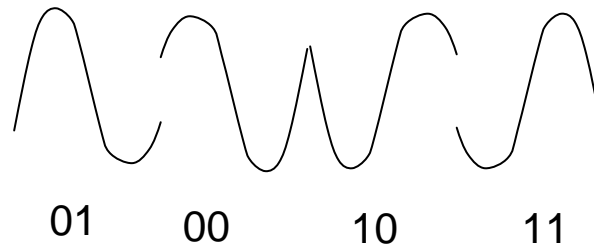


Note this is equivalent to analog QAM if we interpret the first bit and second bit coming from two pulse sequences!

Example of 4-QAM

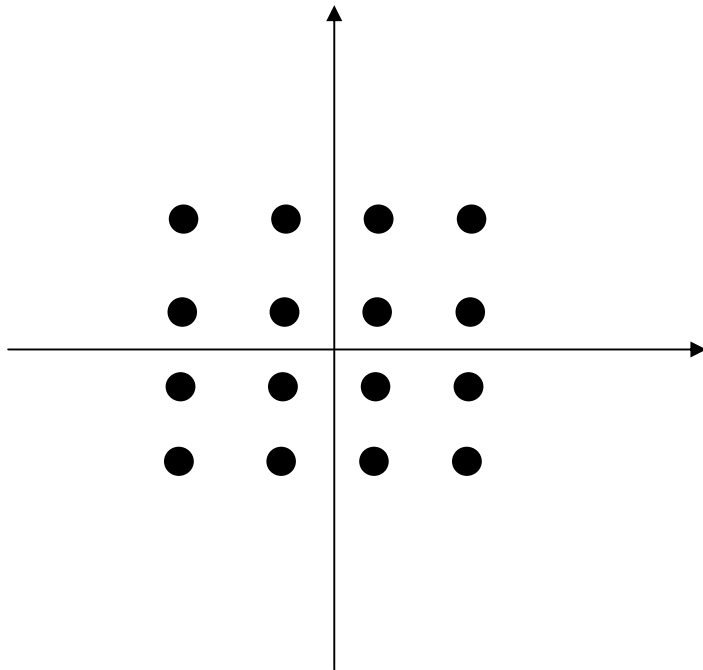
Example: Given a sequence: 01001011..., what is the analog form resulting from 4-ASK?

Using the previous mapping, the analog waveform for the above sequence is

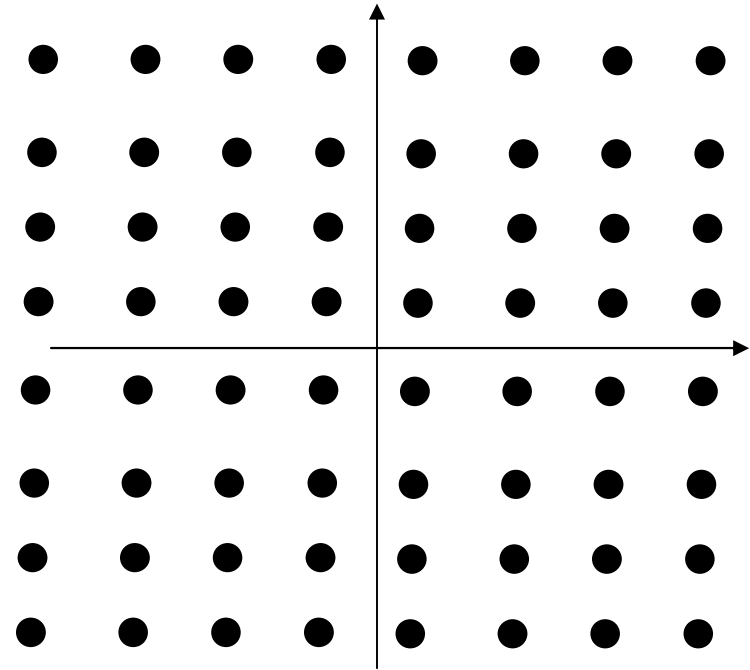


16-QAM, etc.

16 QAM (4 bits/symbol):

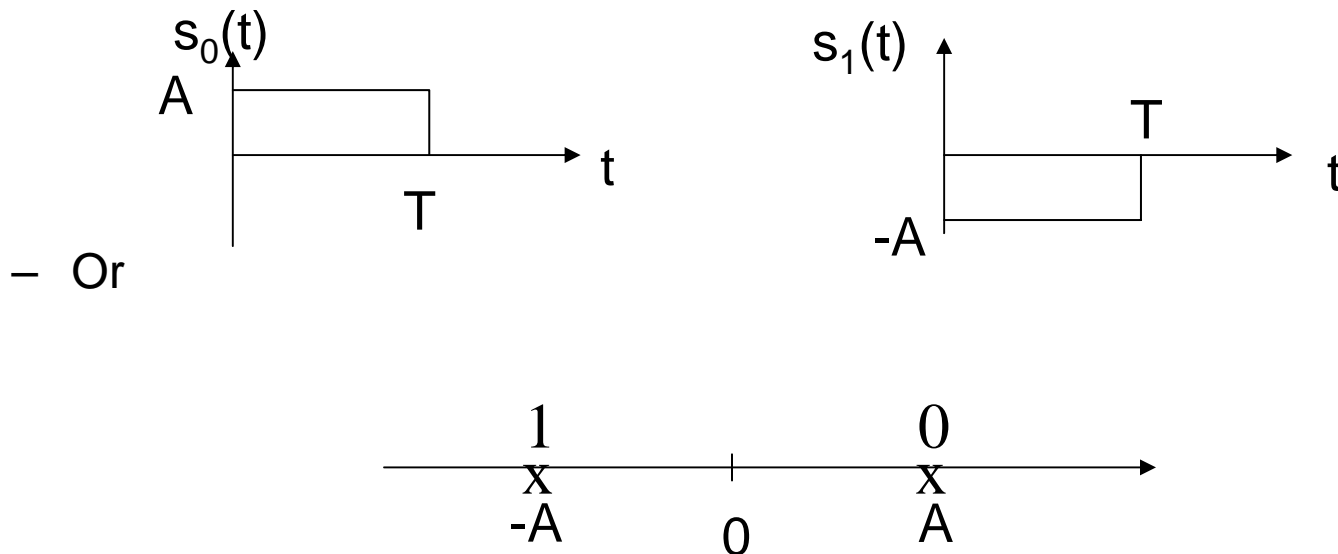


64-QAM (6 bits/symbol)



Vector Representation

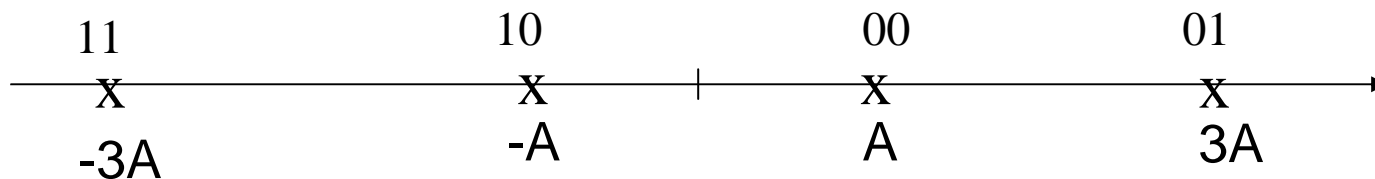
- For BPAM, In $s_0(t)$ and $s_1(t)$ the term $\cos(2\pi f_c t)$ is common
 - We can represent $s_0(t) = A, 0 < t < T$
 $s_1(t) = -A, 0 < t < T$



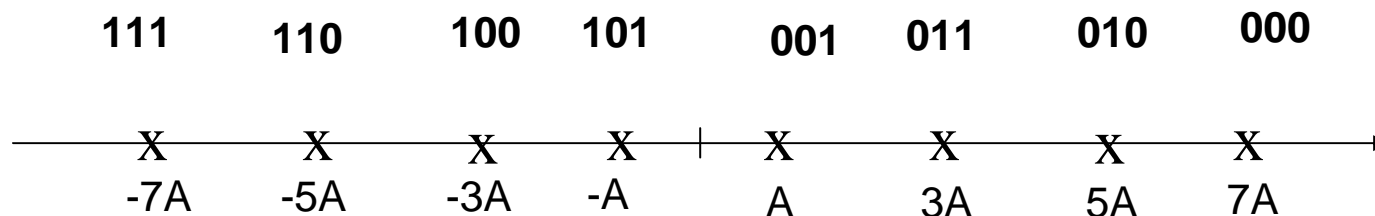
Ex: A digital sequence 100110 is sent as $\{A -A -A A A -A -A\}$

Vector Representation for ASK

- M-ary ASK (M=4, send two bits at a time)



- Labeling is done in such a way that adjacent points only differ in one bit (called Gray mapping)

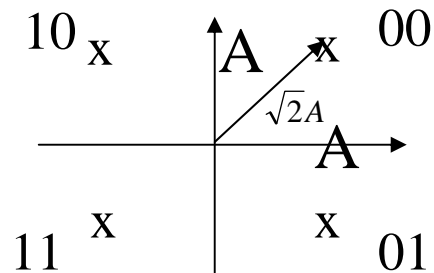


8-ASK

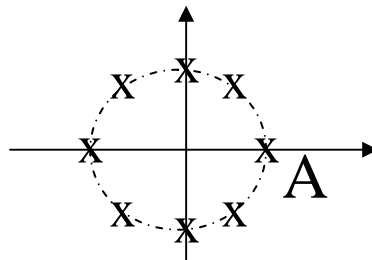
Ex: A digital sequence 100010 is sent as $\{-3A, 5A, \dots\}$

Two Dimensional Modulation

- Modulated signal $s_i(t) = A_c \cos(2\pi f_c t) + A_s \sin(2\pi f_c t)$
 - 4-QAM (Quadrature Amplitude Modulation)



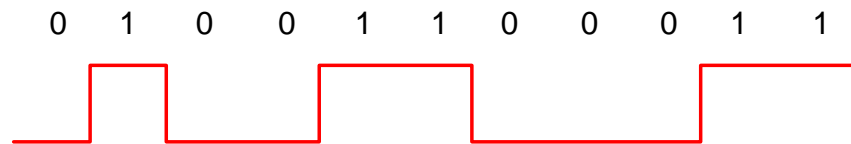
- 8-PSK (Phase Shift Keying)



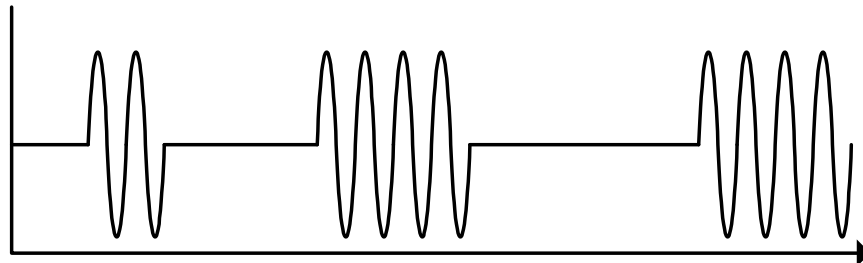
Parameters of Modulation

- Three important parameters of a modulation scheme:
 - Minimum distance d_{\min} : The smallest distance among points in vector representation, which affects error detection capability
 - Average energy E_{av}
 - Number of bits/symbol = $\log_2(M)$
- Example:
 - 4-ASK: $d_{\min}=2A$, $E_{\text{av}}=2(A^2+9A^2)/4=5A^2$, $\log_2(M)=2$
 - 4-QAM: $d_{\min}=2A$, $E_{\text{av}}=2A^2$, $\log_2(M)=2$
 - For the same E_{av} and M , we want to maximize d_{\min} (minimize effect of transmission noise)
 - For the same d_{\min} and M , we want to minimize E_{av} (minimize power consumption)

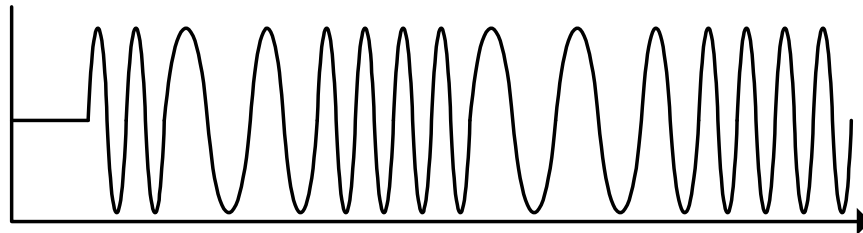
Other Digital Modulation Techniques



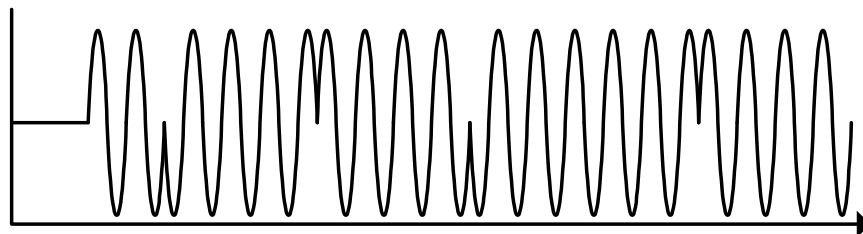
Original signal (in a pulse sequence)



Amplitude shift-keying (ASK)



Frequency shift-keying (FSK)

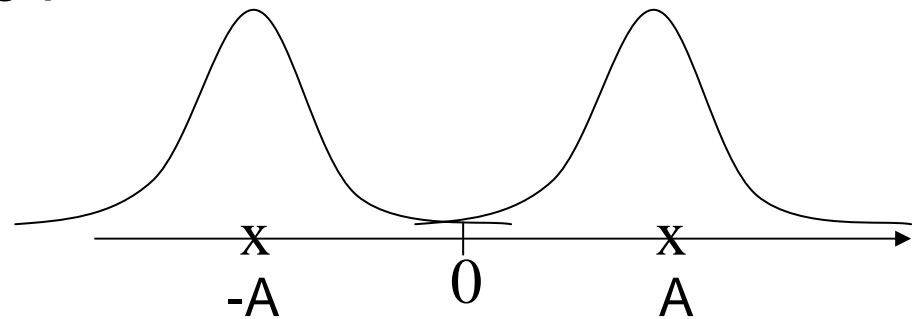


Phase shift-keying (PSK)

Optimum Receiver for AWGN

- Received signal amplitude r

$r=s+n$, $s=A$ or $-A$,
Left is the pdf of r , assuming s
is equally likely to be A and $-A$



- Decide A (“1”) was sent if $r>0$, $-A$ (“0”) was sent if $r<0$.
- How often do we make errors?
 - Depends on the distance $d_{\min}=2A$
 - If $s=A$, n must be $< -d_{\min}/2$
 - If $s=-A$, n must be $> d_{\min}/2$
 - The larger is d_{\min} , the less likely

Error Probability of BPSK

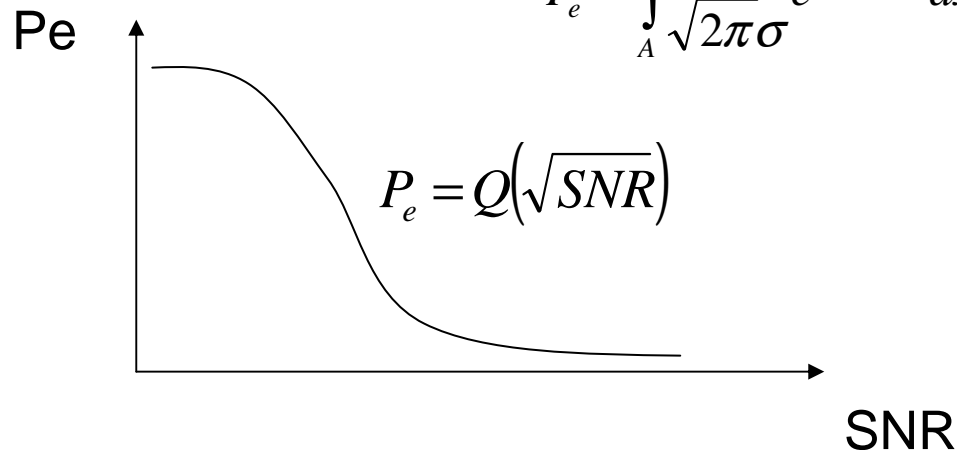
- To have few errors we want A large compared to σ

- Signal to noise ratio= SNR = $\frac{A^2}{\sigma^2}$

- Probability of error $P_e = P(A) \int_{-\infty}^{-A} \frac{1}{\sqrt{2\pi}\sigma} e^{-x^2/2\sigma^2} dx + P(-A) \int_A^{\infty} \frac{1}{\sqrt{2\pi}\sigma} e^{-x^2/2\sigma^2} dx$

If $P(A) = P(-A) = 1/2$

$$P_e = \int_A^{\infty} \frac{1}{\sqrt{2\pi}\sigma} e^{-x^2/2\sigma^2} dx = \int_{A/\sigma}^{\infty} \frac{1}{\sqrt{2\pi}} e^{-x^2/2} dx = Q(\sqrt{SNR})$$



Channel Error Detection

- By adding additional bits to the information bits, we can also detect errors
- Parity Check (check-sum)
 - Append a **parity bit** to the end of each block (called a frame) of k information bits such that the total number of '1' is even or odd
 - Used in IP packet for error detection because of simplicity
- **Example:**
 - ASCII "G" = **1 1 1 0 0 0 1**
 - with even parity = append ?
 - with odd parity = append ?

Error Detection Probability

- Parity check can detect all single bit errors in a block (in fact all error patterns with odd number of error bits)
- Cannot detect double errors or any even number of errors.
- Probability of k error bits in a n -bit frame (assuming bit error rate p_b)

$$p_n(k) = \binom{n}{k} p_b^k (1 - p_b)^{n-k}$$

Ex : if $n = 10^4$, $p_b = 10^{-6}$ (a good channel, e.g. ISDN)

$$p_n(1) \approx 10^{-2}, p_n(2) \approx \frac{1}{2} 10^{-4} \leq p_n(1)$$

if $n = 10^3$, $p_b = 10^{-3}$ (a not so good channel, e.g. wireless)

$$p_n(1) \approx 1, p_n(2) \approx \frac{1}{2}$$

Cyclic-Redundancy Codes (CRC)

General Method:

- The transmitter generates an t -bit check sequence number from a given k -bit data frame such that the resulting $(k+t)$ -bit frame is divisible by some number
- The receiver divides the incoming frame by the same number
- If the result of the division does not leave a remainder, the receiver assumes that there was no error
- If n is large, undetectable error patterns are very unlikely
- Widely used in data communications

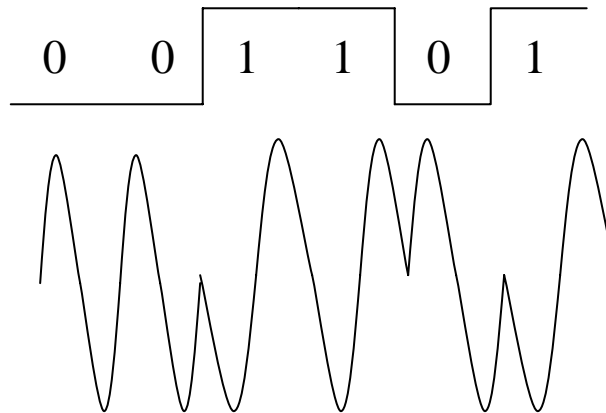
Error Correction by Channel Coding

- In addition to detect errors, we can correct errors by inserting parity bits. This is called **channel coding**.
- Simplest channel code: Repetition coding:
 - Instead of sending one 0, send three 0's.
 - If we receive more 0's than 1's: Decide 000 was sent
 - Error probability=3 or 2 bits are wrong=
 - =probability of receiving 111,110,101,011
 - Smaller than P_e
 - Example: Old $P_e=10^{-2}$, New $P_e=3 \cdot 10^{-4}$
 - But the information rate is decreased
 - Instead of $R=1\text{bit/sec}$, we have $R=1/3\text{ bits/sec}$
- More sophisticated channel codes can correct the same number of errors with lower redundancy (less reduction in information rate)
 - Block codes (Reed-Solomon codes), convolutional codes, turbo codes

$$P_e^3 + 3(1 - P_e)P_e^2 \approx 3P_e^2$$

Channel Capacity

- Channel capacity = maximum number of bits/second we can send for reliable transmission
- What determines this?
 - The receiver typically evaluates the received signal level over the entire bit interval, to determine which bit is sent (0 or 1)
 - If we send more bits/second, the interval of each bit is shorter, channel noise will more easily make the receiver make mistakes
 - With higher order modulation, we decode multiple bits together over each symbol interval



Noiseless Case

- To send a digital sequence, we send a sequence of pulses
 - level A representing “1”, and level $-A$ for “0”
- If the interval of each pulse is T , what is the maximum frequency?
 - The maximum frequency occurs when we have alternating 1 and -1 , spanning $2T$ time per cycle
 - This signal has a maximum frequency of $f_{max} = 1/2T$
 - With a channel with bandwidth B , $f_{max} = 1/2T \leq B$
 - $T \geq 1/2B$, $C = 1/T \leq 2B$
 - we can send at most $C = 2B$ bits/second !
- Instead of two-level pulse, we can use N -level pulse ($m = \log_2 N$ bits/pulse), we can send $C = 2Bm$ bits/second (Nyquist channel capacity)
- By increasing N (hence m), we can reach infinite bit rate!
 - What is wrong?
 - Here we assume the channel is noiseless, every level can be distinguished correctly

Channel Capacity (Noisy Case)

- When there are more levels in a pulse, the signal difference between two adjacent levels is smaller (for the same total dynamic range). Noise in communication channel is more likely to make the received/detected level differ from the actual level.



- The channel capacity depends on the signal to noise ratio (SNR)
 - $SNR = \text{signal energy} / \text{noise energy}$
- Channel Capacity
 - $C = B \log_2 (1 + SNR)$ bits/second
 - Ex: $B = 6\text{MHz}$, $SNR = 20\text{ dB} = 100$, $C = ?$

Advantages of Digital Communication

- More tolerant to channel noise.
 - With amplitude shift keying: as long as the noise does not change the amplitude from one level to another level, the original bits can be inferred
 - With QAM: as long as the received signal is more close to the original symbol than its neighboring symbols
 - Each repeater can regenerate the analog modulated signals from demodulated bits (noise do not accumulate)
- Can insert parity bits before sending data to allow detection/correction of errors
- Can apply digital compression techniques to reduce data rate subject to distortion criterion
- Different signals can be multiplexed more easily
 - Internet packets can contain any types of signals, cell phones can send different types of data

What Should You Know

- What is digital modulation?
- One and two dimensional modulation schemes
- Parameters of digital modulation schemes
 - Can calculate basic parameters for a given modulation scheme
 - Understand the design objective
- Channel error detection and correction
 - Should understand how simple parity check works and how repetition coding works
- Channel capacity
 - Understand the role of channel bandwidth and SNR in determining channel capacity
- Advantages of digital communications

References

- A. Leon-Garcia, I. Widjaja, Communication networks, Chap 3: Digital transmission fundamentals.