Signal Modulation

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Typical Modulation and Coding Schemes in Wireless









Changing signal characteristics including -Phase

- -Amplitude
- -Frequency

Depending on the medium, signal range, and data Properties different encoding techniques can be used

Reasons for Choosing End Modulation Techniques



- Digital data, digital signal
 - Less complex equipments
 - Less expensive than digital-to-analog modulation equipment
- Analog data, digital signal
 - Permits use of modern digital transmission and switching equipment
 - Requires conversion to analog prior to wireless transmission

Reasons for Choosing Encoding Techniques



- Digital data, analog signal Used in Wireless!
 - Some transmission media will only propagate analog signals
 - E.g., optical fiber and unguided media
- Analog data, analog signal
 - Analog data in electrical form can be transmitted easily and cheaply
 - Done with voice transmission over voicegrade lines

A- Refer to notes!

Signal Encoding Criteria

- What determines how successful a receiver will be in interpreting an incoming signal?
 - Signal-to-noise ratio
 Data rate
 Bandwidth
 Signal-to-noise ratio
 1- R ∝ BER
 2- SNR ∝ 1/BER
 3- BW ∝ R
- An increase in data rate <u>increases</u> bit error rate
- An increase in SNR <u>decreases</u> bit error rate
- An increase in bandwidth allows an <u>increase</u> in data rate

Signal Spectrum





Refer to notes!



Question: If data rate is 1 what is the frequency of the carrier in ASK?



Other binary digit represented by absence of carrier

$$s(t) = \begin{cases} A\cos(2\pi f_c t) & \text{binary 1} \\ 0 & \text{binary 0} \end{cases}$$

 \Box where the carrier signal is $A\cos(2\pi f_c t)$



- Susceptible to sudden gain changes
- Inefficient modulation technique
- On voice-grade lines, used up to 1200 bps
- Used to transmit digital data over optical fiber

Binary Frequency-Shift Keying (BFSK)

Two binary digits represented by two different frequencies near the carrier frequency

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$$s(t) = \begin{cases} A\cos(2\pi f_1 t) & \text{binary 1} \\ A\cos(2\pi f_2 t) & \text{binary 0} \end{cases}$$

□ where f_1 and f_2 are offset from carrier frequency f_c by equal but opposite amounts

Binary Frequency-Shift Keying (BFSK)

- Less susceptible to error than ASK
- On voice-grade lines, used up to 1200bps
- Used for high-frequency (3 to 30 MHz) radio transmission
- Can be used at higher frequencies on LANs that use coaxial cable
- Different frequencies can be used to support FULL DUPLEX transmission
 - TX: 1070-1270 Hz
 - RX: 2025-2225 Hz (1200 Hz)

1170 2125

- More than two frequencies are used
- More bandwidth efficient but more susceptible to error

$$s_i(t) = A\cos 2\pi f_i t$$
 $1 \le i \le M$

fm

 $\Box f_i = f_c + (2i - 1 - M)f_d$

Levels

 \Box *f* _c = the carrier frequency

Symbols \Box f_d = the difference frequency (freq. separation)

- \square M = number of different signal elements = 2^L
 - \Box L = number of bits per signal element

Example:

- Assume fc = 250 Khz
- □ Required frequency separation is 25 KHz (fd=25KHz)
- \square 8 levels of signals (M=8)
- Use Multi-FSK
- Answer the following questions:
 - 1. What is the center frequency of the modulated signal?
 - 2. How many different frequencies do we need for this system?
 - 3. How many bits do we need to generate the data?
 - 4. What are M different frequencies?
 - 5. What is the period of each symbol (How many bits per symbol)?
 - 6. What is the BW of each symbol?
 - 7. What is the total BW required?
 - 8. What is the data rate (total BW/bit)?

Fc=250Khz / 8 diff. freq. / 3 bits/ f1 to f8:75K(000),125(001), 175(101) .../3 bits per symbol / (next slide)

Example:



To match data rate of input bit stream, each output signal element is held for:

 $T_s = LT$ seconds

 \Box where T is the bit period (data rate = 1/T)

So, one signal element encodes L bits



Total bandwidth required

 (# of symbols x BW/symbol)=2Mf_d

 Minimum frequency separation
 required BW/symbol =2f_d=1/T_s

Therefore, modulator requires a bandwidth of

$$W_d = 2^L/LT = M/T_s$$





Multiple Frequency-Shift Keying (MFSK) - Example





Phase-Shift Keying (PSK)



Phase-Shift Keying (PSK) -Variations

0

0

1

1

0

Differential PSK (DPSK)

- Phase shift with reference to previous bit
- Binary 0 signal burst of same phase as previous signal burst
- Binary 1 signal burst of opposite phase to previous signal burst

1

0

0

0

1

NRZ-L*: $0 \rightarrow +V$; $1 \rightarrow -V$

* Nonreturn zero, level

Differential Phase Shift Keying (DPSK) 0 - no phase change

1 - phase change





QPSK and OQPSK Modulators





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Phase change never exceeds 180 deg! \rightarrow Large phase shift is hard to implement in HW (imaging going from $00 \rightarrow 11$)



11

p/4

3p/4

Phase change never exceeds 90 deg!

OQPSK Modulator - Example



11

- p/4

p/4

3p/4

Phase change never exceeds 90 deg!

So what is the difference in performance?QPSK and OQPSK Modulators

- Both have the same spectral characteristics and error performance
- Max. phase change for QPSK is 180 deg.
- Max. phase change for OQPSK is 90 deg.
 - Results in smaller sudden phase change → good for limiting non-linearity impact
 - Less non-linearity → less signal spread → less interference

Phase-Shift Keying (PSK)

Multilevel-PSK

- Using multiple phase angles with each angle having more than one amplitude, multiple signals elements can be achieved
- Example: Standard 9600 baud used in Modem
 - □ 12 phase angles / four of them have 2 diff. amplitude levels → 16 levels



8-QPSK (same amplitude!)

Modulation Impact on Performance

Modulation rate
 Bit Error Ratio (rate)
 Bit Energy Level
 Bandwidth Efficiency

Modulation Impact on Performance

$$D = \frac{R}{L} = \frac{R}{\log_2 M}$$

 \square *D* = modulation rate, baud

 \square *R* = data rate = 1/Tb, bps

- \square M = number of different signal elements = 2^{L}
- \Box L = number of bits per signal element

Example: We can support a data rate of 9600 bps Using 2400 baud rate if M = 16, L = 4 using a complex modulation scheme!

This is how we can transmit more bits in the same medium!



Performance



 $Eb/No=(S/R)/No; N=NoB_T; \rightarrow Eb/No(dB) = S/N(dB) - R/B_T (dB)$

Eb/No is the ratio of energy per bit to noise power density per hertz

Performance Comparison



Bandwidth Efficiency is proportional to BER

Performance

□ Bandwidth of modulated signal (B_T)
 ■ ASK/PSK/FSK

MPSK

MFSK

notes

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- L = number of bits encoded per signal element
- M = number of different signal elements
- r <1; a constant; depends on filtering

 $B_T = (1+r)R$ $B_T = \left(\frac{1+r}{L}\right)R = \left(\frac{1+r}{\log_2 M}\right)R$

 $B_T = \left(\frac{(1+r)M}{\log_2 M}\right)R$

BW Efficiency = R/BT = Modulation Rate/BT

Remember: Larger number of states \rightarrow higher data rate \rightarrow Higher potential error!

Example

Find BW_efficiency for PSK if BER 10^{-7} with SNR = 12 dB

Eb/No = 11.2 from figure \rightarrow **Eb/No(dB)** = **S/N(dB)** - **R/B_T (dB)** R/BT = BW_efficiency (dB) = 12 - 11.2 = 0.8 R/BT = 1.2



Reasons for Analog Modulation

Modulation of digital signals

- When only analog transmission facilities are available, digital to analog conversion required
- Modulation of analog signals
 - A higher frequency may be needed for effective transmission
 - Modulation permits frequency division multiplexing



Basic Modulation Techniques

- Analog data to analog signal
 - Amplitude modulation (AM)
 - Angle modulation
 - Frequency modulation (FM)
 - Phase modulation (PM)

http://mason.gmu.edu/~mlyons3/AM_FM/AM_FM_model.html

Amplitude modulation



Frequency modulation (FM)

With frequency modulation, the modulating signal and the carrier are combined in such a way that causes the carrier FREQUENCY (fc) to vary above and below its normal (idling) frequency
 As the voltage of the modulating signal increases in the positive direction from A to B, the frequency of the carrier is increased in proportion to the to the modulating voltage
 The amplitude of the carrier remains constant



Phase modulation (PM)

The phase of the carrier is changed by the change in amplitude of the modulating signal The modulated carrier wave is lagging the carrier wave when the modulating frequency is positive (A and B) When the modulating frequency is negative, the modulated carrier wave is leading the carrier wave (C and D)



Varying the phase of the carries linearly in proportion to the modulating signal such that maximum phase shift occurs during positive and negative peaks of the modulating signal.

Comparison

FM and PM require greater bandwidth than AM

Applet: http://engweb.info/courses/wdt/lecture07/wdt07-am-fm.html#AM_Applet_

Basic Encoding Techniques

- Analog data to digital signal
 - Pulse code modulation (PCM)
 - Delta modulation (DM)
- Basic process of digitizing analog data



The question is how to represent the digital data



Pulse Code Modulation

- Based on the sampling theorem
- Each analog sample is assigned a binary code
 - Analog samples are referred to as pulse amplitude modulation (PAM) samples
- The digital signal consists of block of *n* bits, where each *n*-bit number is the amplitude of a PCM pulse

Pulse Code Modulation



Sampling frequency (two times fmax) Quantization levels (number of bits available)

http://www.netbook.cs.purdue.edu/animations/convert%20analog%20to%20digital.html

Pulse Code Modulation

- By quantizing the PAM pulse, original signal is only approximated
 - More quantization levels \rightarrow more accurate signal approximation \rightarrow more complex system
- Leads to quantizing noise
- Signal-to-noise ratio for quantizing noise

 $SNR_{dB} = 20 \log 2^{n} + 1.76 dB = 6.02n + 1.76 dB$

NOTE: each additional bit increases SNR by 6 dB, or a factor of 4

Example:

- Assuming we use 7 bits to reconstruct the voice signal. Bandwidth of voice signal is 4KHz.
 - How may quantization levels can we create?
 - What is the sampling rate for the voice signal?
 - What is the BW of the PCM-encoded digital signal?
 - What is the minimum BW required using the Nyquist criterion?
 - How much the s/N (in dB) will increase if we use 9 bits instead?

Example:

Assuming we use 7 bits to reconstruct the voice signal. Bandwidth of voice signal is

- 4KHz. 2^{^7} = 128 levels
 - How Sampling rate: 2B = 8KHz (8000 samples / sec) ← according to the sampling theorem What Each sample has 7 bits
 - What PCM BW = 8000 sample/sec x 7 bit/ sample = 56 bit/sec
- reate? e signal? ligital
- What is the minimum BW required using the Nyquist criterion?
- How much the s/N (in dB) will increase if we use 9 bits instead?

Delta Modulation

- Analog input is approximated by staircase function
 - Moves up or down by one quantization level (δ) at each sampling interval
- Only the change of information is sent
 - only an increase or decrease of the signal amplitude from the previous sample is sent
 - a no-change condition causes the modulated signal to remain at the same 0 or 1 state of the previous sample

Delta Modulation

- Two important parameters
 - Size of step assigned to each binary digit (δ)
 - Sampling rate
- Accuracy improved by increasing sampling rate
 - However, this increases the data rate
- Advantage of DM over PCM is the simplicity of its implementation

References

- Applets
 - PM and FM Applet:
 - <u>http://cnyack.homestead.com/files/modulation/modfmpm.htm</u>
 - <u>http://williams.comp.ncat.edu/Networks/modulate.htm</u>
 - □ Very good:
 - http://sem.mosaic-service.com/electron2/frequency_conversion.htm
 - Learn about sampling theorem: <u>http://www.facstaff.bucknell.edu/mastascu/elessonshtml/Signal/SignalNoteNyquistSampling.htm</u>
- Very good basic information about AM, FM: <u>http://cbdd.wsu.edu/kewlcontent/cdoutput/TR502/page21.htm</u>
- Read about The Nyquist Sampling Theorem www.facstaff.bucknell.edu/mastascu/elessonshtml/Signal/ SignalNoteNyquistSampling.htm