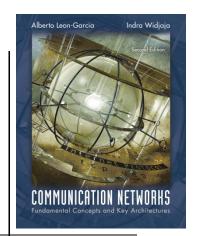
Chapter 3 Digital Transmission Fundamentals



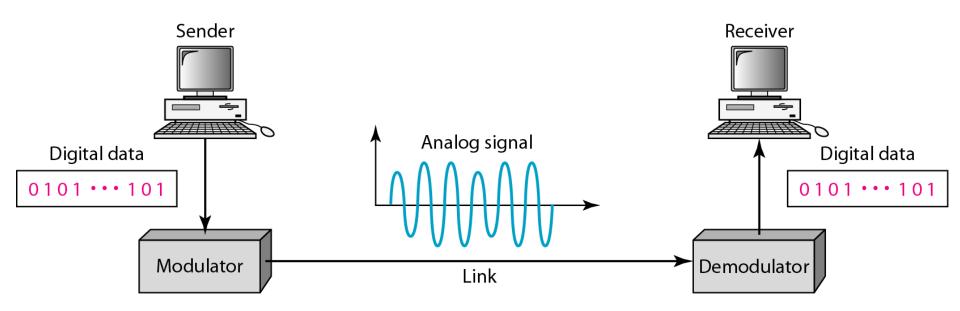
Modems and Digital Modulation

CSE 3213, Winter 2010
Instructor: Foroohar Foroozan



Modulation of Digital Data

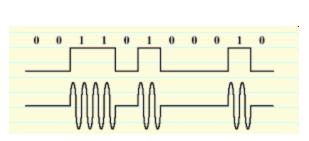




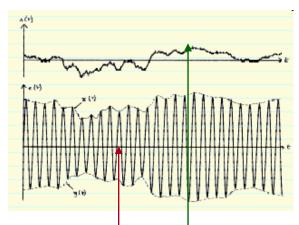
Modulation of Digital Data

Modulation

 process of converting digital data or a low-pass analog signal to band-pass (higher-frequency) analog signal

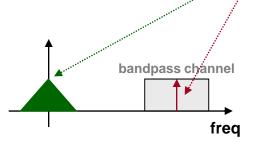


Digital-to-analog modulation.



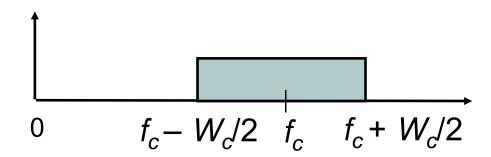
Anal<mark>og-to-analog modulation.</mark>

- aka carrier frequency modulated signal signal high frequency signal that acts as a basis for the information signal
 - information signal is called modulating signal



Bandpass Channels

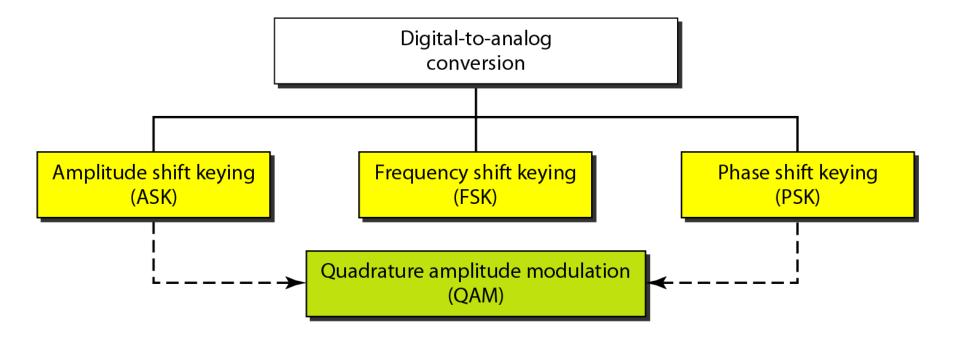




- Bandpass channels pass a range of frequencies around some center frequency f_c
 - Radio channels, telephone & DSL modems
- Digital modulators embed information into waveform with frequencies passed by bandpass channel
- Sinusoid of frequency f_c is centered in middle of bandpass channel
- Modulators embed information into a sinusoid

Digital To Analog Modulation

- process of changing one of the characteristics of an analog signal (typically a sinewave) based on the information in a digital signal
 - sinewave is defined by three characteristics (<u>amplitude</u>, <u>frequency</u>, and <u>phase</u>) ⇒ digital data (binary 0 and 1) can be represented by varying any of the three
 - application: transmission of digital data over telephone wire (modem)

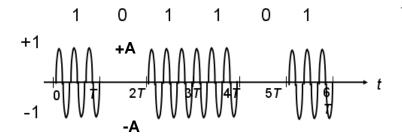


Amplitude Modulation



- **ASK** strength of the carrier signal is varied to represent binary 1 or 0
 - both frequency and phase remain constant while the amplitude changes
 - commonly, one of the amplitudes is zero

$$s(t) = \begin{cases} A_0 cos(2\pi f_c t), & binary 0 \\ A_1 cos(2\pi f_c t), & binary 1 \end{cases} = \begin{cases} 0, & binary 0 \\ Acos(2\pi f_c t), & binary 1 \end{cases}$$

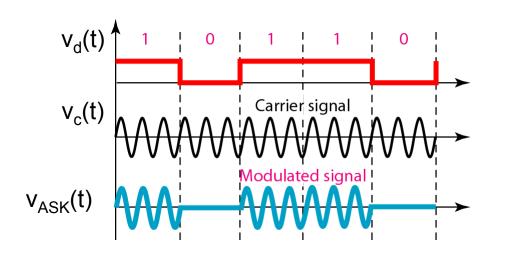


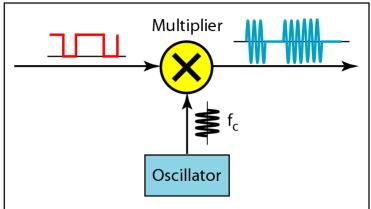
- demodulation: only the presence or absence of a sinusoid in a given time interval needs to be determined
- advantage: simplicity
- disadvantage: ASK is very susceptible to noise interference <u>noise</u> usually (only) affects the amplitude, therefore ASK is the modulation technique most affected by noise
- application: ASK is used to transmit digital data over optical fiber

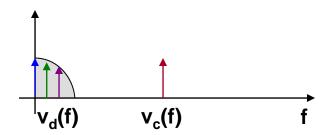
Amplitude Modulation (Cont.)

Example [ASK]









Amplitude Modulation (Cont.)



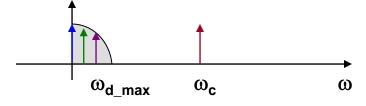
ASK-Modulated Signal: Frequency Spectrum

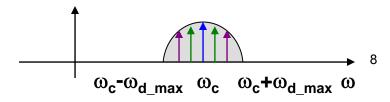
$$\cos A \cdot \cos B = \frac{1}{2} (\cos(A - B) + \cos(A + B))$$

Carrier signal:
$$v_c(t) = cos(2\pi f_c t) = cos(\omega_c t)$$
, where $2\pi f_c = \omega_c$

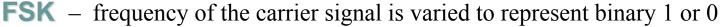
Digital signal:
$$v_d(t) = A \cdot \left[\frac{1}{2} + \frac{2}{\pi} \cos \omega_0 t - \frac{2}{3\pi} \cos 3\omega_0 t + \frac{2}{5\pi} \cos 5\omega_0 t - \dots \right]$$

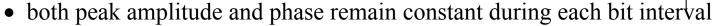
$$\begin{aligned} & \text{Modulated signal:} \quad v_{\text{ASK}}(t) = v_{\text{c}}(t) \cdot v_{\text{d}}(t) = \\ & = cos\omega_{\text{c}}t \cdot \left[\frac{1}{2} + \frac{2}{\pi}cos\omega_{\text{o}}t - \frac{2}{3\pi}cos3\omega_{\text{o}}t + \frac{2}{5\pi}cos5\omega_{\text{o}}t - ...\right] = \\ & = \frac{1}{2}cos\omega_{\text{c}}t + \frac{2}{\pi}cos\omega_{\text{c}}t \cdot cos\omega_{\text{o}}t - \frac{2}{3\pi}cos\omega_{\text{c}}t \cdot cos3\omega_{\text{o}}t + ... = \\ & = \frac{1}{2}cos\omega_{\text{c}}t + \frac{1}{\pi}\left[cos(\omega_{\text{c}} - \omega_{\text{o}})t + cos(\omega_{\text{c}} + \omega_{\text{o}})t\right] - \\ & - \frac{1}{3\pi}\left[cos(\omega_{\text{c}} - 3\omega_{\text{o}})t + cos(\omega_{\text{c}} + 3\omega_{\text{o}})t\right] + ... \end{aligned}$$

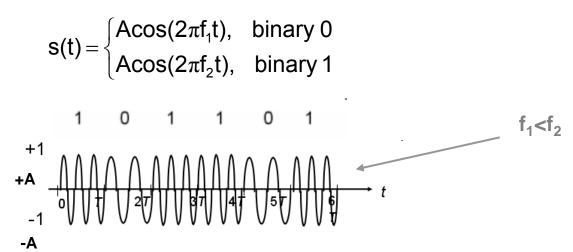




Frequency Modulation





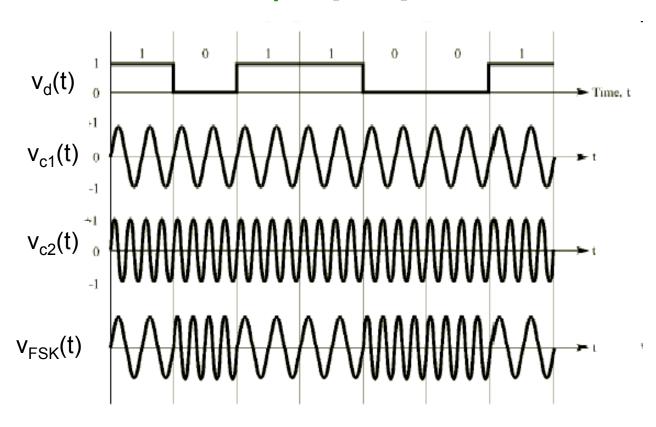


- demodulation: demodulator must be able to determine which of two possible frequencies is present at a given time
- advantage: FSK is less susceptible to errors than ASK receiver is looking for specific frequency changes over a number of intervals, so voltage (noise) spikes can be ignored
- disadvantage: FSK spectrum is 2 x ASK spectrum
- application: over voice lines, in high-frequency radio transmission, etc.

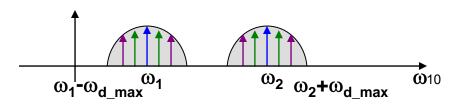
Frequency Modulation (Cont.)



Example [FSK]







Frequency Modulation (Cont.)

FSK-Modulated Signal: Frequency Spectrum



Digital signal: $V_d(t)$ - modulated with ω_1 , and

 $v_d'(t) = 1 - v_d(t)$ - modulated with ω_2

$$\begin{split} \text{Modulated signal:} \quad & v_{\text{FSK}}(t) = cos\omega_1 t \cdot v_d(t) + cos\omega_2 t \cdot (1 - v_d(t)) = \\ & = cos\omega_1 t \cdot \left[\frac{1}{2} + \frac{2}{\pi} cos\omega_0 t - \frac{2}{3\pi} cos3\omega_0 t + \frac{2}{5\pi} cos5\omega_0 t - ... \right] + \\ & + cos\omega_2 t \cdot \left[\frac{1}{2} - \frac{2}{\pi} cos\omega_0 t + \frac{2}{3\pi} cos3\omega_0 t - \frac{2}{5\pi} cos5\omega_0 t - ... \right] = \\ & = ... \\ & = \frac{1}{2} cos\omega_1 t + \frac{1}{\pi} \left[cos(\omega_1 - \omega_0) t + cos(\omega_1 + \omega_0) t \right] - \\ & - \frac{1}{3\pi} \left[cos(\omega_1 - 3\omega_0) t + cos(\omega_1 + 3\omega_0) t \right] + ... + \\ & \frac{1}{2} cos\omega_2 t - \frac{1}{\pi} \left[cos(\omega_2 - \omega_0) t + cos(\omega_2 + \omega_0) t \right] - \\ & + \frac{1}{3\pi} \left[cos(\omega_2 - 3\omega_0) t + cos(\omega_2 + 3\omega_0) t \right] + ... + \end{split}$$

Phase Modulation



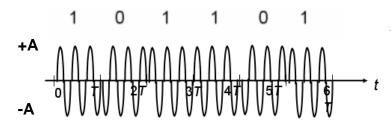
PSK – phase of the carrier signal is varied to represent binary 1 or 0

- peak amplitude and frequency remain constant during each bit interval
- <u>example</u>: binary 1 is represented with a phase of 0° , while binary 0 is represented with a phase of 180° = π rad \Rightarrow PSK is equivalent to multiplying the carrier signal by +1 when the information is 1, and by

-1 when the information is 0

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

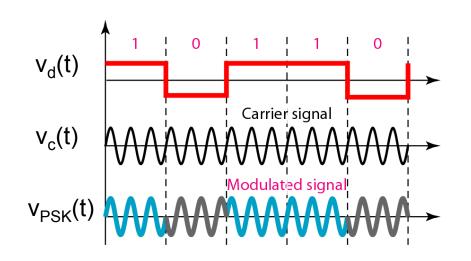
$$s(t) = \begin{cases} A\cos(2\pi f_c t), & \text{binary 1} \\ -A\cos(2\pi f_c t), & \text{binary 0} \end{cases}$$

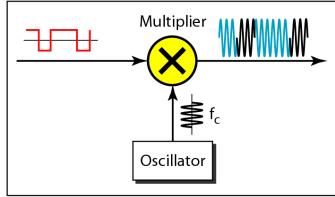


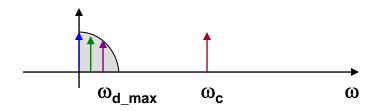
- demodulation: demodulator must be able to determine the phase of received sinusoid with respect to some reference phase
- advantage: PSK is less susceptible to errors than ASK, while it requires/occupies the same bandwidth as ASK
 - more efficient use of bandwidth (higher data-rate) are possible, compared to FSK !!!
- disadvantage: more complex signal detection / recovery process, than in ASK and FSK

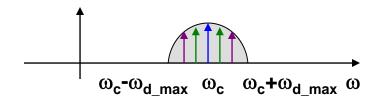
Example [PSK]













$$\cos^2 A = \frac{1}{2} \left(1 + \cos 2A \right)$$

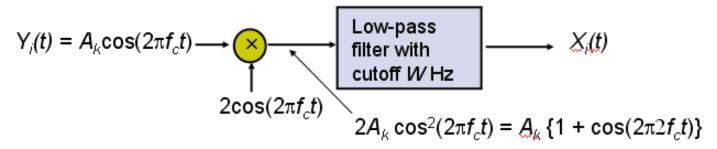
PSK Detection / Recovery – multiply the received / modulated signal $\pm A\cos(2\pi f_c t)$ by $\frac{2*\cos(2\pi f_c t)}{2}$

resulting signal

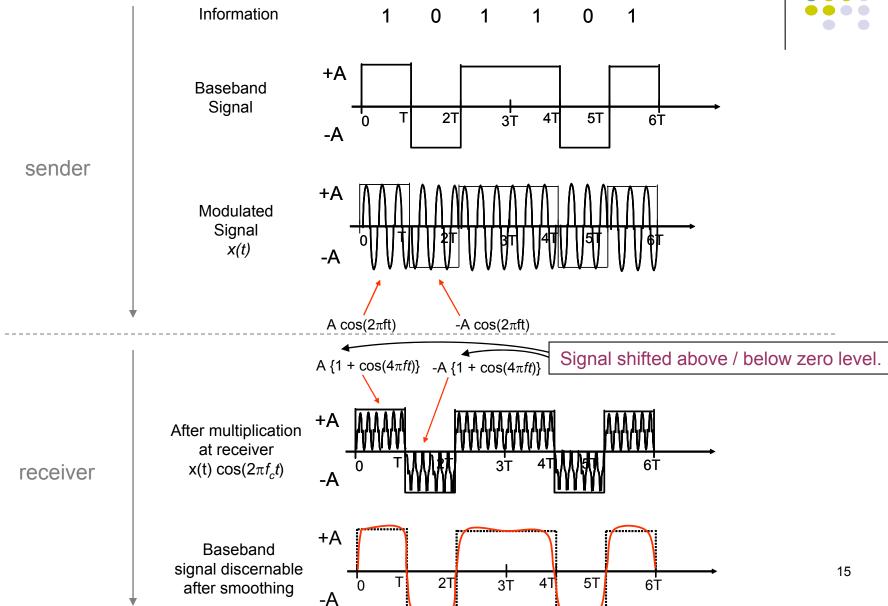
$$2A\cos^{2}(2\pi f_{c}t) = A[1+\cos(4\pi f_{c}t)], \text{ binary 1}$$

$$-2A\cos^{2}(2\pi f_{c}t) = -A[1+\cos(4\pi f_{c}t)], \text{ binary } 0$$

• by removing the oscillatory part with a low-pass filter, the original baseband signal (i.e. the original binary sequence) can be easily determined







Signaling rate and Transmission Bandwidth



Fact from modulation theory:

If Baseband signal x(t) with bandwidth B Hz then B Modulated signal $x(t)\cos(2\pi f_c t)$ has bandwidth 2B Hz $f_c - B \qquad f_c \qquad f_c + B$

- If bandpass channel has bandwidth W_c Hz,
 - Then baseband channel has $W_c/2$ Hz available, so
 - modulation system supports $W_c/2 \times 2 = W_c$ pulses/second

Recall baseband transmission system of bandwidth W_c [Hz] can theoretically support 2 W_c pulses/sec

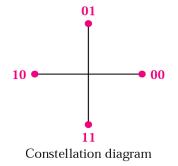


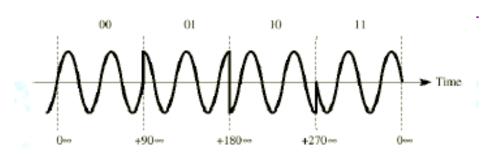
QPSK = 4-PSK

- PSK that uses phase shifts of $90^{\circ}=\pi/2$ rad \Rightarrow 4 different signals are generated, each representing 2 bits

$$s(t) = \begin{cases} Acos(2\pi f_c t), & binary \ 00 \\ Acos(2\pi f_c t + \frac{\pi}{2}), & binary \ 01 \\ Acos(2\pi f_c t + \pi), & binary \ 10 \\ Acos(2\pi f_c t + \frac{3\pi}{2}), & binary \ 11 \end{cases}$$

Dibit	Phase
00	0
01	90
10	180
11	270
Dibit (2 bits)	





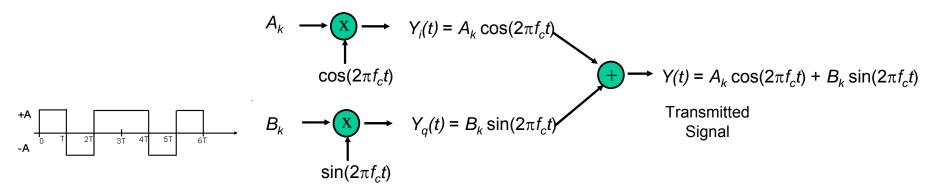
- advantage: higher data rate than in PSK (2 bits per bit interval), while bandwidth occupancy remains the same
- 4-PSK can easily be extended to 8-PSK, i.e. n-PSK
- however, higher rate PSK schemes are limited by the ability of equipment to distinguish small differences in phase

Quadrature Amplitude Modulation

(QAM)

- uses "two-dimensional" signalling
 - original information stream is <u>split into two sequences</u> that consist of odd and even symbols, e.g. B_k and A_k

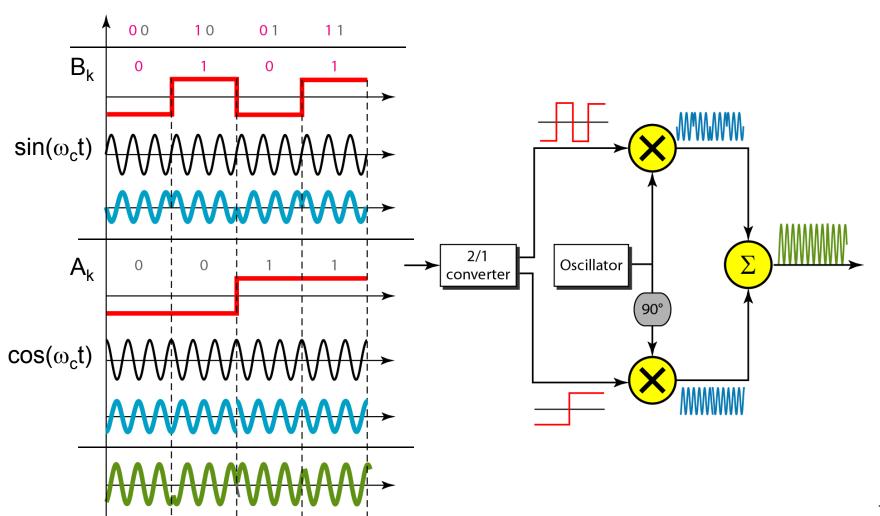
- A_k sequence (in-phase component) is modulated by $\cos(2\pi f_c t)$, while B_k sequence (quadrature-phase component) is modulated by $\sin(2\pi f_c t)$
- composite signal $A_k \cos(2\pi f_c t) + B_k \sin(2\pi f_c t)$ is sent through the channel



advantage: data rate = 2 bits per bit-interval!

QAM (Cont.)

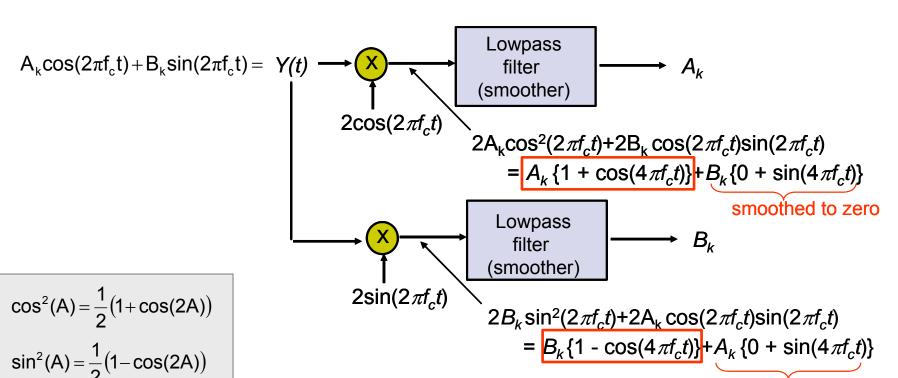
Example [QAM]



QAM (Cont.)

QAM Demodulation

- by multiplying Y(t) by $2 \cdot \cos(2\pi f_c t)$ and then low-pass filtering the resultant signal, sequence A_k is obtained
- by multiplying Y(t) by $2 \cdot \sin(2\pi f_c t)$ and then low-pass filtering the resultant signal, sequence B_k is obtained



sin(2A) = 2sin(A)cos(A)

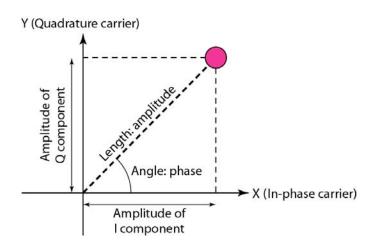
smoothed to zero

Signal Constellations

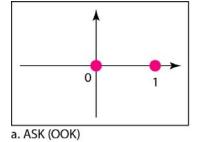
Constellation Diagram

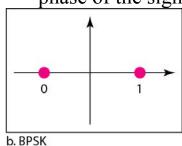
 used to represents possible symbols that may be selected by a given modulation scheme as points in 2-D plane

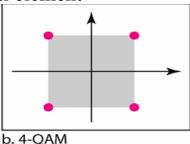




- X-axis is related to in-phase carrier: $cos(\omega_c t)$
 - the projection of the point on the X-axis defines the peak amplitude of the in-phase component
- Y-axis is related to the quadrature carrier: $\sin(\omega_c t)$
 - the projection of the point on the Y-axis defines the peak amplitude of the quadrature component
- the length of the line that connects the point to the origin is the peak amplitude of the signal element (combination of X and Y components)
- the angle the line makes with the X-axis is the phase of the signal element



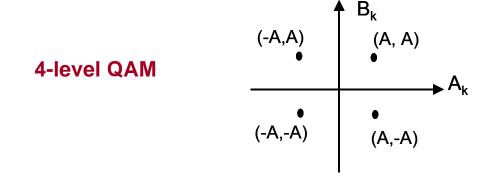




QAM (Cont.)

QAM cont. – QAM can also be seen as a combination of ASK and PSK

$$Y(t) = A_k \cos(2\pi f_c t) + B_k \sin(2\pi f_c t) = \left(A_k^2 + B_k^2\right)^{\frac{1}{2}} \cos(2\pi f_c t + \tan^{-1}\frac{B_k}{A_k})$$



QAM (Cont.)



16-level QAM – the number of bits transmitted per T [sec] interval can be further increased by increasing the number of levels used

- in case of 16-level QAM, A_k and B_k individually can assume 4 different levels: -1, -1/3, 1/3, 1
- data rate: 4 bits/pulse ⇒ 4W bits/second

$$Y(t) = A_k cos(2\pi f_c t) + B_k sin(2\pi f_c t) = \left(A_k^2 + B_k^2\right)^{\frac{1}{2}} cos(2\pi f_c t + tan^{-1} \frac{B_k}{A_k})$$

$$A_k = A_k and B_k individually can take on 4 different values; the resultant signal can take on (only) 3 different values!!!$$

In QAM various combinations of amplitude and phase are employed to achieve higher digital data rates.

Amplitude changes are susceptible to noise \Rightarrow the number of phase shifts used by a QAM system is always greater than the number of amplitude shifts.