

Ex. For a standard telephone circuit with a signal-to-noise power ratio of 1000 (30 dB) and a bandwidth of 2.7 kHz, the Shannon limit for information capacity is?

M-ary Encoding

M-ary is a term derived from the word *binary*

$$N = \log_2 M$$

$$2^N = M$$

Where N = number of bits necessary

M = number of conditions, levels, or combinations

Baud and Minimum Bandwidth

Baud refers to the rate of change of a signal on the transmission medium after encoding and modulation have occurred.

$$\text{Baud} = 1/T_s$$

where baud = symbol rate (baud per second)

T_s = time of one signaling element (seconds)

Minimum Nyquist Bandwidth - the minimum theoretical bandwidth necessary to propagate a signal

$$f_b = B \log_2 M \quad \text{where } f_b = \text{channel capacity (bps)}$$

B = minimum Nyquist bandwidth (hertz)

M = number of discrete signal or voltage levels

$$B = \frac{f_b}{\log_2 M} = \frac{f_b}{N}$$

The **Baud** and the **ideal minimum Nyquist bandwidth** have the same value and are equal to the bit rate divided by the number of bits encoded.

AMPLITUDE-SHIFT KEYING

The simplest digital modulation technique where a binary information signal directly modulates the amplitude of an analog carrier. Sometimes called *digital amplitude modulation* (DAM).

Mathematically:

$$v_{(ask)}(t) = [1 + v_m(t)] \left[\frac{A}{2} \cos(\omega_c t) \right]$$

Where $v_{ask}(t)$ = amplitude-shift keying wave

$v_m(t)$ = digital information (modulating) signal (volts)

$A/2$ = unmodulated carrier amplitude (volts)

ω_c = analog carrier radian frequency (radians per second, $2\pi f_c t$)

The modulating signal $[v_m(t)]$ is a normalized binary waveform;

where +1 V = logic 1

-1 V = logic 0.

Therefore, for a logic 1 input, $v_m(t) = +1$ V, substituting

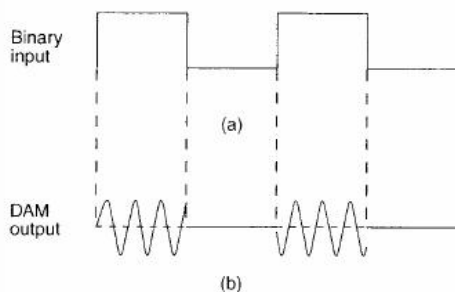
$$v_{(ask)}(t) = [1 + 1] \left[\frac{A}{2} \cos(\omega_c t) \right]$$
$$= A \cos(\omega_c t)$$

and for a logic 0 input, $v_m(t) = -1$ V, substituting

$$v_{(ask)}(t) = [1 - 1] \left[\frac{A}{2} \cos(\omega_c t) \right]$$

Thus, the modulated wave $v_{ask}(t)$, is either $A \cos(\omega_c t)$ = "on" or 0 = "off"

The waveform Digital Amplitude Modulation



-for every change in the input binary data stream, there is one change in the ASK waveform

- The rate of change of the ASK waveform (baud) is the same as the rate of change of the binary input (bps)

$$B = f_b / 1 = f_b \quad \text{baud} = f_b / 1 = f_b$$

Ex. Determine the baud and minimum nyquist bandwidth necessary to pass a 10kbps binary signal using ASK, for ASK, $N = 1$?

FREQUENCY-SHIFT KEYING

- is a form of constant-amplitude angle modulation similar to standard frequency modulation (FM) except the modulating signal is a binary signal that varies between two discrete voltage levels. Sometimes called *binary FSK* (BFSK)

Mathematically: $v_{fsk}(t) = V_c \cos[2\pi[f_c + v_m(t) \Delta f]t]$

Where

$v_{fsk}(t)$ = binary FSK waveform

V_c = peak analog carrier amplitude (volts)

f_c = analog carrier center frequency (hertz)

Δf = peak change (shift) in the analog carrier frequency (hertz)

$v_m(t)$ = binary input (modulating) signal (volts)

The modulating signal is a normalized binary waveform

where logic 1 = +1 V

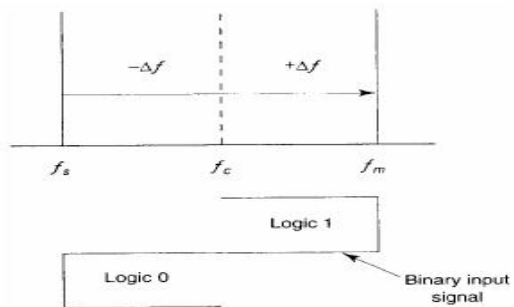
logic 0 = -1 V.

Thus, for a logic 1 input, $v_m(t) = +1$

for a logic 0 input, $v_m(t) = -1$

$$v_{fsk}(t) = V_c \cos[2\pi(f_c + \Delta f)t]$$

$$v_{fsk}(t) = V_c \cos[2\pi(f_c - \Delta f)t]$$



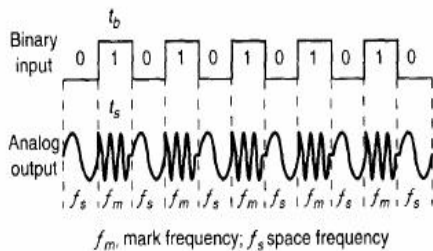
- the carrier center frequency (f_c) is shifted (deviated) up and down in the frequency domain

Frequency deviation is expressed mathematically:

$$\Delta f = |f_m - f_s| / 2 \quad \text{Where: } \Delta f = \text{frequency deviation (hertz)}$$

$|f_m - f_s|$ = absolute difference between the mark and space frequencies (hertz)

$f_m = f_{\text{USB}}$; $f_s = f_{\text{LSB}}$



(a)

binary input	frequency output
0	space (f_s)
1	mark (f_m)

(b)

FSK Bit Rate, Baud, and Bandwidth

The baud for binary FSK can also be determined by substituting $N = 1$

$$\text{baud} = f_b / 1 = f_b$$

The minimum bandwidth for FSK is given as

$$B = |(f_s - f_b) - (f_m - f_b)|$$

$$= |(f_s - f_m)| + 2f_b$$

$$B = 2(\Delta f + f_b)$$

FSK in the time domain: (a) waveform: (b) truth table

Example:

Determine (a) the peak frequency deviation, (b) minimum bandwidth, and (c) baud for a binary FSK signal with a mark frequency of 49 kHz, a space frequency of 51 kHz, and an input bitrate of 2 kbps.

$$f_a = f_b / 2$$

where f_a = highest fundamental frequency of the binary input signal (hertz)

f_b = input bit rate (bps)

The formula used for modulation index in FM is also valid for FSK, so

$$mf = \Delta f / f_a$$

substituting:

$$mf = |(f_m - f_s) / 2| / f_b / 2$$

$$mf = |f_m - f_s| / f_b$$

Using a Bessel table, determine the minimum bandwidth for the same FSK signal with a mark frequency of 49 kHz, a space frequency of 51 kHz, and an input bit rate of 2 kbps.

PHASE-SHIFT KEYING - another form of *angle-modulated, constant-amplitude* digital modulation.

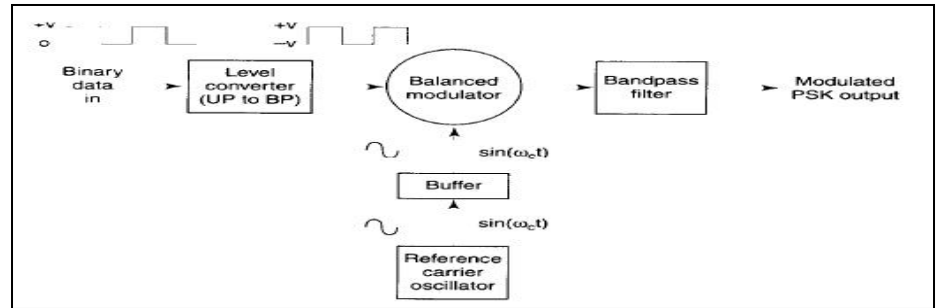
Types: a) BPSK b) QPSK c) 8PSK d) 16PSK

A) The simplest form of PSK is *binary phase-shift keying (BPSK)*, where $N = 1$ and $M = 2$. Therefore, with BPSK, two phases ($2^1 = 2$) are possible for the carrier.

There are two possible phases, Logic 1 and logic 0, the phase of the output carrier shifts between two angles that are separated by 180° .

BPSK TRANSMITTER:

The balanced modulator acts as a phase reversing switch. Depending on the logic condition of the digital input, the carrier is transferred to the output either in phase or 180° out of phase with the reference carrier oscillator.



(a) Balanced ring modulator;

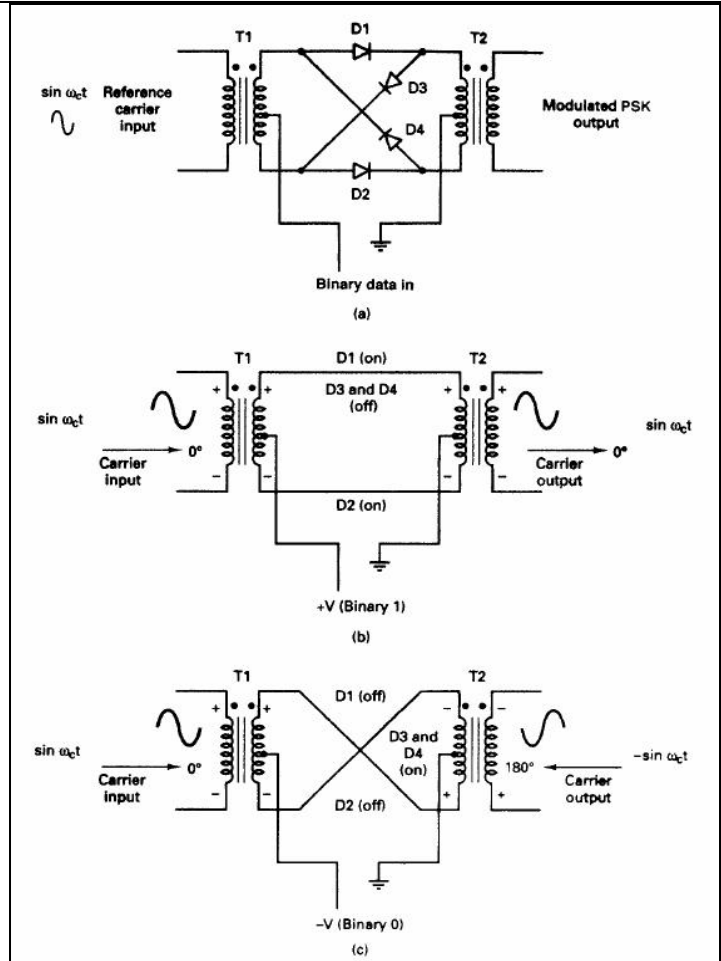
For the balanced modulator to operate properly, the digital input voltage must be much greater than the peak carrier voltage (V_c).

(b) logic 1 input;

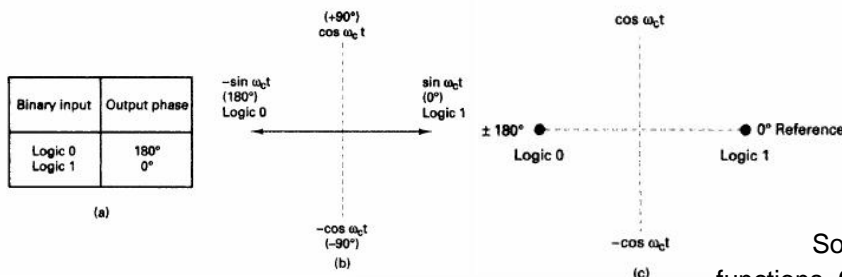
If the binary input is a logic 1 (positive voltage), diodes D1 and D2 are forward biased and on, while diodes D3 and D4 are reverse biased and off. With the polarities shown, the carrier voltage is developed across transformer T2 in phase with the carrier voltage across T1. Consequently, the output signal is in phase with the reference oscillator.

(c) logic 0 input

If the binary input is a logic 0 (negative voltage), diodes D1 and D2 are reverse biased and off, while diodes D3 and D4 are forward biased and on. As a result, the carrier voltage is developed across transformer T2 180° out of phase with the carrier voltage across T1.



BPSK modulator: (a) truth table; (b) phasor diagram; (c) constellation diagram



Mathematically, BPSK modulator output:

$$\text{BPSK output} = [\sin(2\pi f_a t)] \times [\sin(2\pi f_c t)]$$

Where: f_a = maximum fundamental frequency of binary input (hertz)
 f_c = reference carrier frequency (hertz)

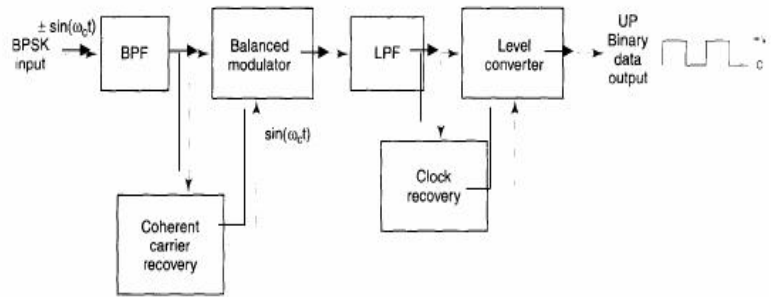
Solving for the trig identity for the product of two sine functions, $0.5 \cos[2\pi(f_c - f_a)t] - 0.5 \cos[2\pi(f_c + f_a)t]$

Bandwidth of BPSK:

Example: For a BPSK modulator with a carrier frequency of 70 MHz and an input bit rate of 10 Mbps, determine the maximum and minimum upper and lower side frequencies, draw the output spectrum, determine the minimum Nyquist bandwidth, and calculate the baud.

BPSK RECEIVER

- The input signal maybe $+\sin \omega_c t$ or $-\sin \omega_c t$.
- The balanced modulator is a product detector; the output is the product of the two inputs (the BPSK signal and the recovered carrier)



Mathematically, the demodulation process is as follows.

For a BPSK input signal of $+\sin \omega_c t$ (logic 1), the output of the balanced modulator is

$$\text{output} = (\sin \omega_c t)(\sin \omega_c t) = \sin^2 \omega_c t$$

Or

$$\sin^2 \omega_c t = 0.5(1 - \cos 2\omega_c t) = 0.5 - 0.5\cos 2\omega_c t$$

filtered out

leaving output = $+ 0.5 \text{ V} = \text{logic 1}$

For a BPSK input signal of $-\sin \omega_c t$ (logic 0), the output of the balanced modulator is

$$\text{output} = (-\sin \omega_c t)(\sin \omega_c t) = -\sin^2 \omega_c t$$

or

$$-\sin^2 \omega_c t = -0.5(1 - \cos 2\omega_c t) = -0.5 + 0.5\cos 2\omega_c t$$

filtered out

leaving

output = $- 0.5 \text{ V} = \text{logic 0}$

Quaternary Phase-Shift Keying (QPSK) is an M-ary encoding scheme where $N = 2$ and $M = 4$. Produce four different input combinations, : 00, 01, 10, and 11. The binary input data are combined into groups of two bits, called *dibits*. Each dibit code generates one of the four possible output phases ($+45^\circ, +135^\circ, -45^\circ$, and -135°).

QPSK transmitter

where logic 1 = +1 V
logic 0 = -1 V

I balanced modulator ($+\sin \omega_c t, -\sin \omega_c t$)

Q balanced modulator ($+\cos \omega_c t, -\cos \omega_c t$)

The output of the linear summer:

QI@00

$I = -1 \sin \omega_c t$

$Q = -1 \cos \omega_c t$

QI@01

$I = +1 \sin \omega_c t$

$Q = -1 \cos \omega_c t$

QI@10

$I = -1 \sin \omega_c t$

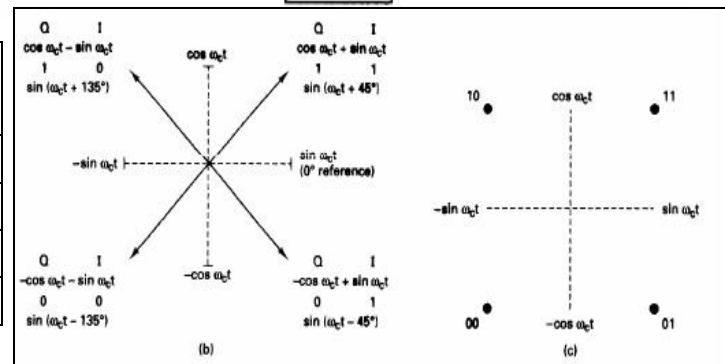
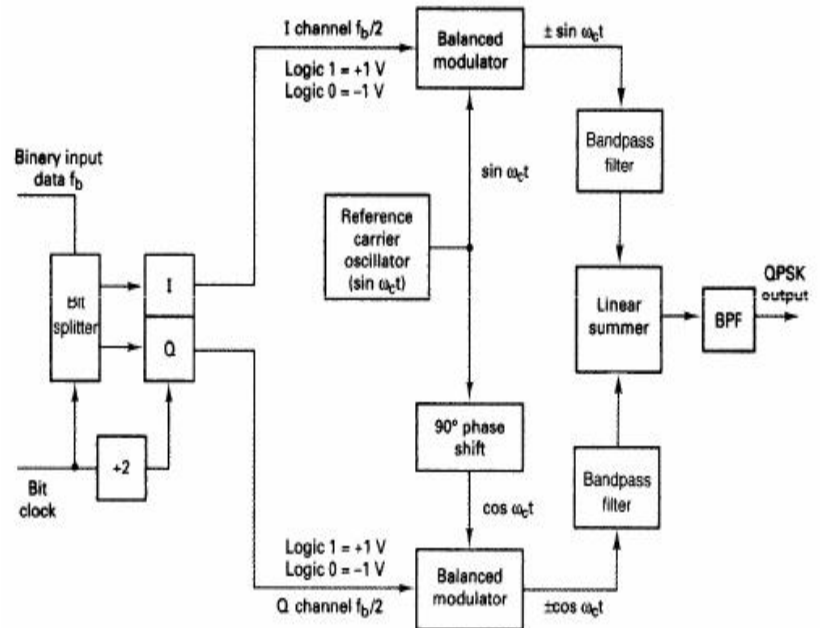
$Q = +1 \cos \omega_c t$

QI@11

$I = +1 \sin \omega_c t$

$Q = +1 \cos \omega_c t$

Binary Input		Amplitude	Output Phase
Q	I		
0	0	1.414	
0	1	1.414	
1	0	1.414	
1	1	1.414	



QPSK receiver

Four possible input signal

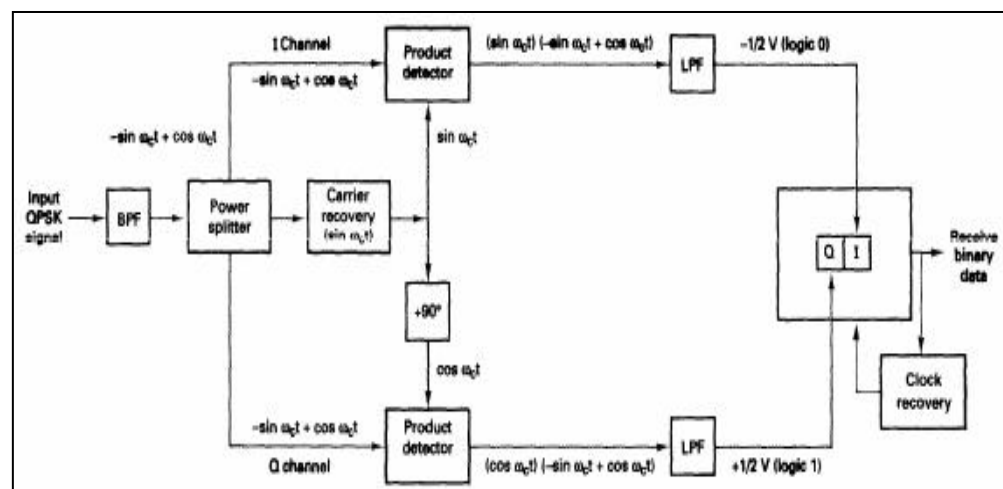
$-1 \sin \omega_c t -1 \cos \omega_c t$

$+1 \sin \omega_c t -1 \cos \omega_c t$

$-1 \sin \omega_c t +1 \cos \omega_c t$

$+1 \sin \omega_c t +1 \cos \omega_c t$

@I channel



$$I = \underbrace{(-\sin \omega_c t + \cos \omega_c t)}_{\text{QPSK input signal}} \underbrace{(\sin \omega_c t)}_{\text{carrier}}$$

$$= (-\sin \omega_c t)(\sin \omega_c t) + (\cos \omega_c t)(\sin \omega_c t)$$

$$= -\sin^2 \omega_c t + (\cos \omega_c t)(\sin \omega_c t)$$

$$= -\frac{1}{2}(1 - \cos 2\omega_c t) + \frac{1}{2}\sin(\omega_c + \omega_c)t + \frac{1}{2}\sin(\omega_c - \omega_c)t$$

$$I = -\frac{1}{2} + \frac{1}{2}\cos 2\omega_c t + \frac{1}{2}\sin 2\omega_c t + \frac{1}{2}\sin 0$$

$$= -\frac{1}{2}V(\text{logic 0})$$

@Q channel

$$Q = \underbrace{(-\sin \omega_c t + \cos \omega_c t)}_{\text{QPSK input signal}} \underbrace{(\cos \omega_c t)}_{\text{carrier}}$$

$$= \cos^2 \omega_c t - (\sin \omega_c t)(\cos \omega_c t)$$

$$= \frac{1}{2}(1 + \cos 2\omega_c t) - \frac{1}{2}\sin(\omega_c + \omega_c)t - \frac{1}{2}\sin(\omega_c - \omega_c)t$$

$$Q = \frac{1}{2} + \frac{1}{2}\cos 2\omega_c t - \frac{1}{2}\sin 2\omega_c t - \frac{1}{2}\sin 0$$

$$= \frac{1}{2}V(\text{logic 1})$$

8-PSK

With 8-PSK, the incoming bits are encoded in groups of three, called tribits ($2^3 = 8$), producing eight different input combinations: 000, 001, 010, 011, 100, 101, 110, 111 and producing eight different output phase ($\pm 22.5^\circ, \pm 67.5^\circ, \pm 112.5^\circ, \pm 157.5^\circ$).

8-PSK transmitter

I	C	Output
0	0	-0.541 V
0	1	-1.307 V
1	0	+0.541 V
1	1	+1.307 V

(a)

Q	\bar{C}	Output
0	1	-1.307 V
0	0	-0.541 V
1	1	+1.307 V
1	0	+0.541 V

(b)

The I or Q bit determines the polarity of the output analog

Signal: logic 1=+V and logic 0 = -V

The C determines the magnitude

logic 1= 1.307 V and logic 0 =0.541 V

The output of the linear summer:

QIC@000

IC = -0.541 $\sin \omega_c t$

QC = -1.307 $\cos \omega_c t$

QIC@001

IC = -1.307 $\sin \omega_c t$

QC = -0.541 $\cos \omega_c t$

QIC@010

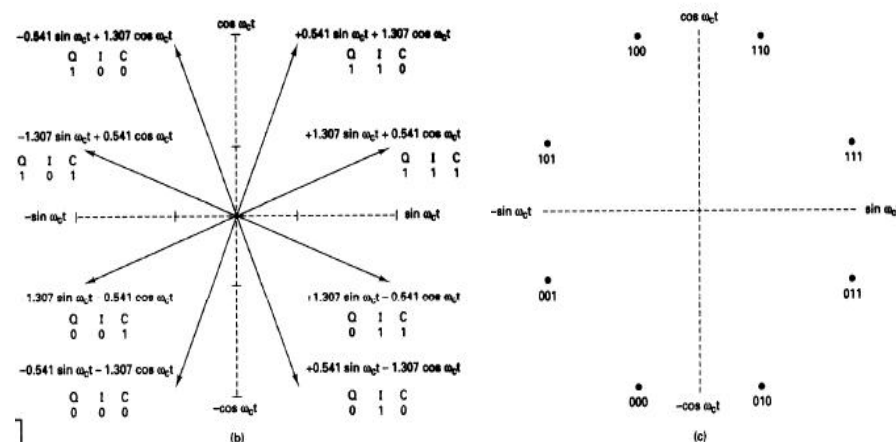
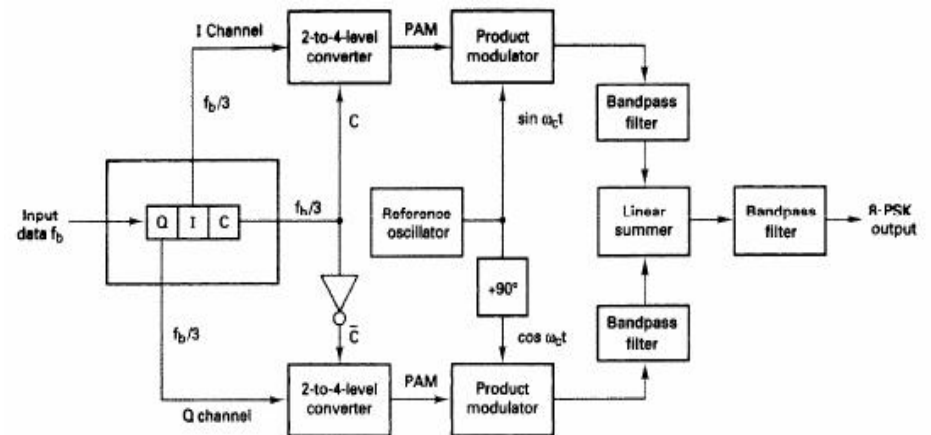
IC = +0.541 $\sin \omega_c t$

QC = -1.307 $\cos \omega_c t$

QIC@011

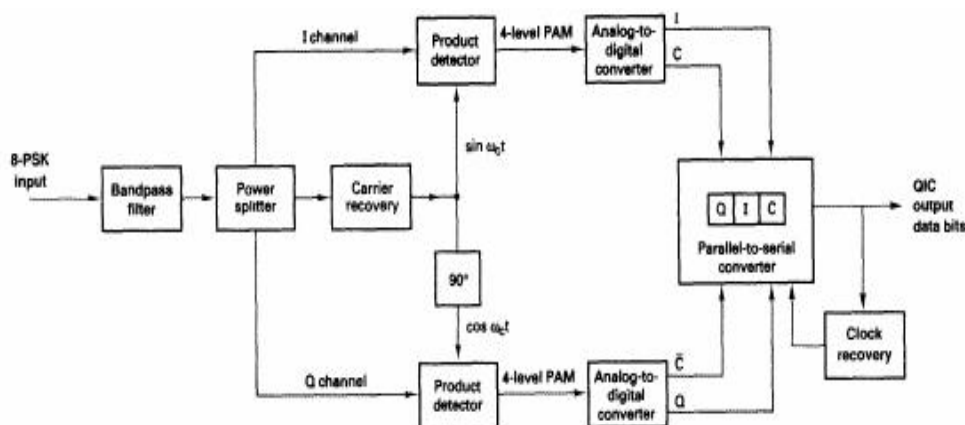
IC = +1.307 $\sin \omega_c t$

QC = -0.541 $\cos \omega_c t$



Q	I	C	Output phase
0	0	0	
0	0	1	
0	1	0	
0	1	1	
1	0	0	
1	0	1	
1	1	0	
1	1	1	

8-PSK receiver

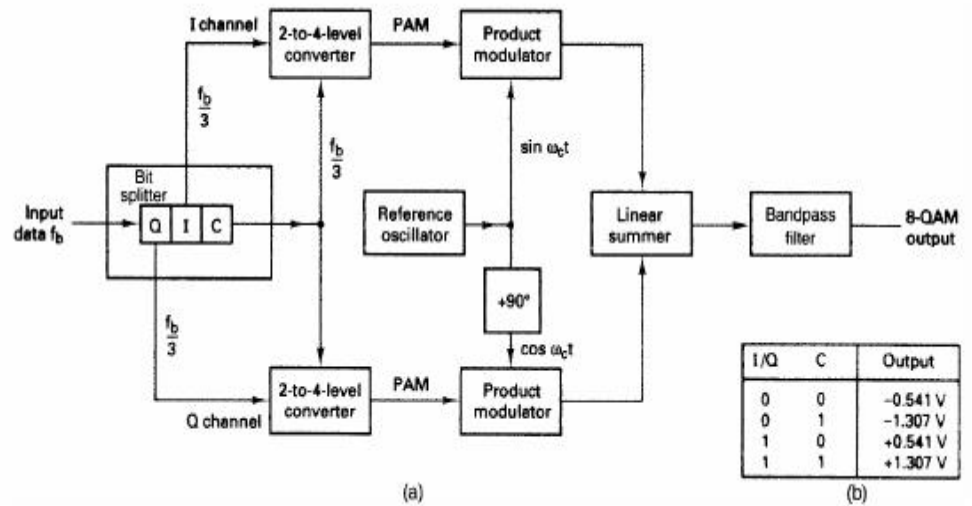


QUADRATURE – AMPLITUDE MODULATION

8-QAM is an M-ary encoding technique where $M = 8$. Unlike 8-PSK, the output signal from an 8-QAM modulator is not a constant-amplitude signal. There are two output amplitudes, and only four phases are possible ($+45^\circ, +135^\circ, -45^\circ$, and -135°)

8-QAM transmitter

Q	I	C	amplitude	Output phase
0	0	0		
0	0	1		
0	1	0		
0	1	1		
1	0	0		
1	0	1		
1	1	0		
1	1	1		



An 8-QAM receiver is almost identical to the 8-PSK receiver

16-QAM

16-QAM is an M-ary system where $M = 16$. The input data are acted on in groups of four ($2^4 = 16$). Both the phase and the amplitude of the transmit carrier are varied. Listed are possible output phases ($\pm 15^\circ, \pm 45^\circ, \pm 75^\circ, \pm 105^\circ, \pm 135^\circ, \pm 165^\circ$)

The I and Q bits determine the polarity
 logic 1 = positive
 logic 0 = negative

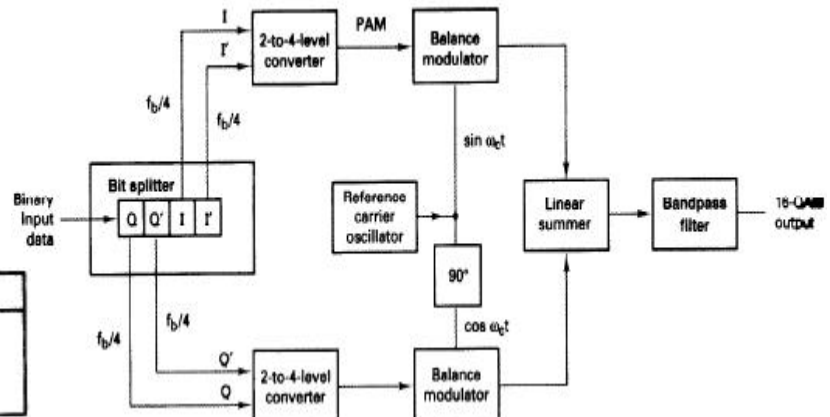
The I' and Q' determine the magnitude
 logic 1 = 0.821 V
 logic 0 = 0.22 V

I	I'	Output
0	0	-0.22 V
0	1	-0.821 V
1	0	+0.22 V
1	1	+0.821 V

(a)

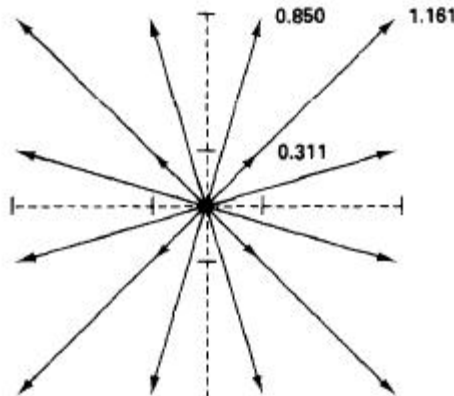
Q	Q'	Output
0	0	-0.22 V
0	1	-0.821 V
1	0	+0.22 V
1	1	+0.821 V

(b)

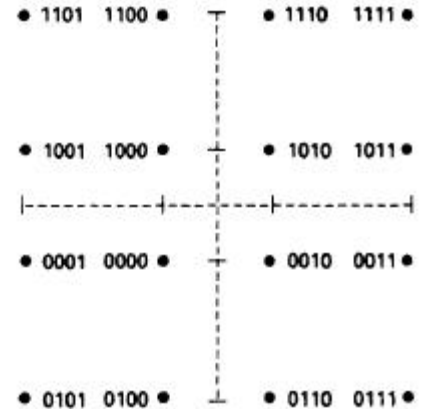


Binary input				16-QAM output	
Q	Q'	I	I'		
0	0	0	0	0.311 V	-135°
0	0	0	1	0.850 V	-165°
0	0	1	0	0.311 V	-45°
0	0	1	1	0.850 V	-15°
0	1	0	0	0.850 V	-105°
0	1	0	1	1.161 V	-135°
0	1	1	0	0.850 V	-75°
0	1	1	1	1.161 V	-45°
1	0	0	0	0.311 V	135°
1	0	0	1	0.850 V	165°
1	0	1	0	0.311 V	45°
1	0	1	1	0.850 V	15°
1	1	0	0	0.850 V	105°
1	1	0	1	1.161 V	135°
1	1	1	0	0.850 V	75°
1	1	1	1	1.161 V	45°

(a)



(b)



(c)

ASK, FSK, PSK and QAM summary

Modulation	Encoding Scheme	Outputs Possible	Minimum Bandwidth	Baud
ASK	Single bit	2	f_b	f_b
FSK	Single bit	2	f_b	f_b
BPSK	Single bit	2	f_b	f_b
QPSK	Dibits	4	$f_b / 2$	$f_b / 2$
8-PSK	Tribits	8	$f_b / 3$	$f_b / 3$
8-QAM	Tribits	8	$f_b / 3$	$f_b / 3$
16-PSK	Quadbits	16	$f_b / 4$	$f_b / 4$
16-QAM	Quadbits	16	$f_b / 4$	$f_b / 4$