

Digital modulation

Lecture 4

Digital modulation

- Digital modulation is the process of changing one of the characteristics of an analog signal based on the information in digital data.
 - ✓ This is also known as “Digital-to-Analog Conversion”
- Topics discussed in this section:
 - Aspects of Digital-to-Analog Conversion
 - Amplitude Shift Keying
 - Frequency Shift Keying
 - Phase Shift Keying
 - Quadrature Amplitude Modulation

Digital modulation

- Digital modulation is the process of changing one of the characteristics of an analog signal based on the information in digital data.

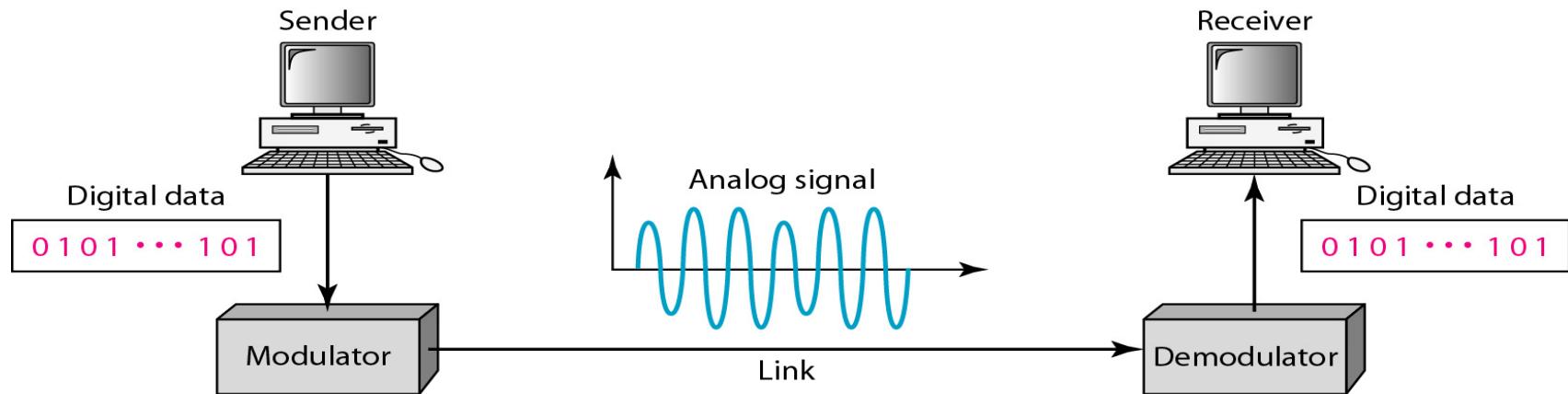
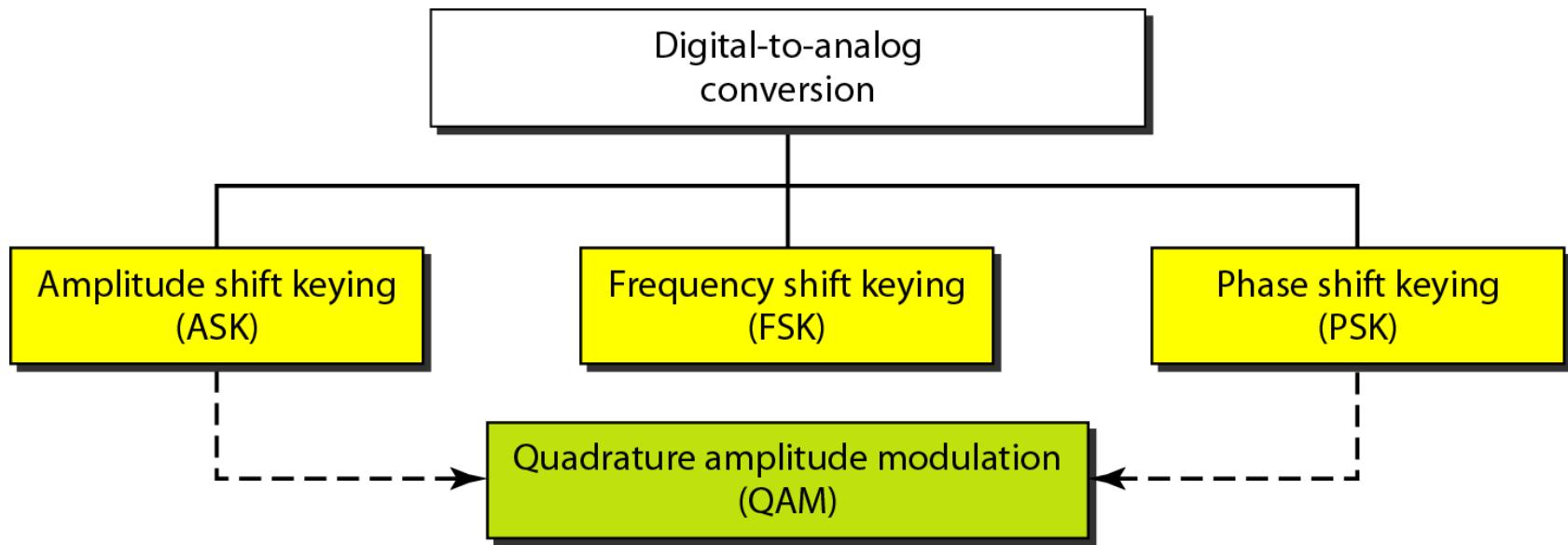


Fig. 1: Digital modulation

- Type of various Digital modulation techniques are:
 - Amplitude Shift Keying
 - Frequency Shift Keying
 - Phase Shift Keying
 - Quadrature Amplitude Modulation

Digital-to-Analog Conversion

- Types of Digital-to-Analog conversion:



Digital-to-Analog Conversion

- Some Aspects of Digital-to-Analog Conversion:
 - Bit Rate: Bit rate is the **number of bits** per second.
 - Baud Rate: Baud rate is the **number of signal elements** per second.
- In the analog transmission of digital data, the baud rate is **less than or equal** to the bit rate.

Digital-to-Analog Conversion

- Example 1:

An analog signal carries **4 bits** per **signal element**. If **1000 signal elements** are sent per second, find the **bit rate**.

- Solution:

In this case, $r = 4$, $S = 1000$, and N is unknown. We can find the value of N from

$$S = N \times \frac{1}{r} \quad \text{or} \quad N = S \times r = 1000 \times 4 = 4000 \text{ bps}$$

- Example 2:

An analog signal has a **bit rate** of **8000 bps** and a **baud rate** of **1000 baud**. How many **data elements** are carried by each signal element? How many **signal elements** do we need?

- Solution:

In this example, $S = 1000$, $N = 8000$, and r and L are unknown. We find first the value of r and then the value of L .

$$\begin{aligned} S &= N \times \frac{1}{r} \quad \rightarrow \quad r = \frac{N}{S} = \frac{8000}{1000} = 8 \text{ bits/baud} \\ r &= \log_2 L \quad \rightarrow \quad L = 2^r = 2^8 = 256 \end{aligned}$$

Amplitude Shift Keying (ASK)

- Amplitude Shift Keying (ASK):

In Amplitude Shift Keying, the **amplitude** of the carrier signal is **varied** to create signal elements. Both **frequency** and **phase** remain **constant** while the **amplitude changes**.

- Binary amplitude shift keying:

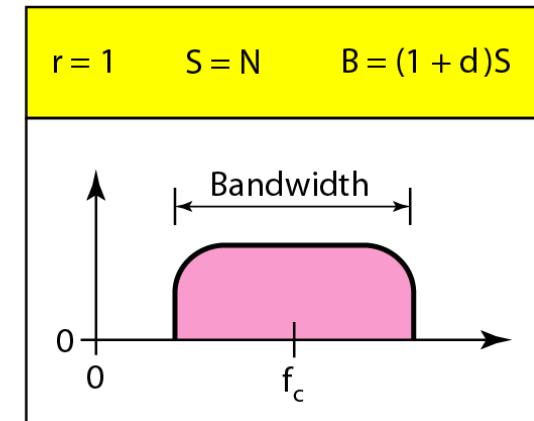
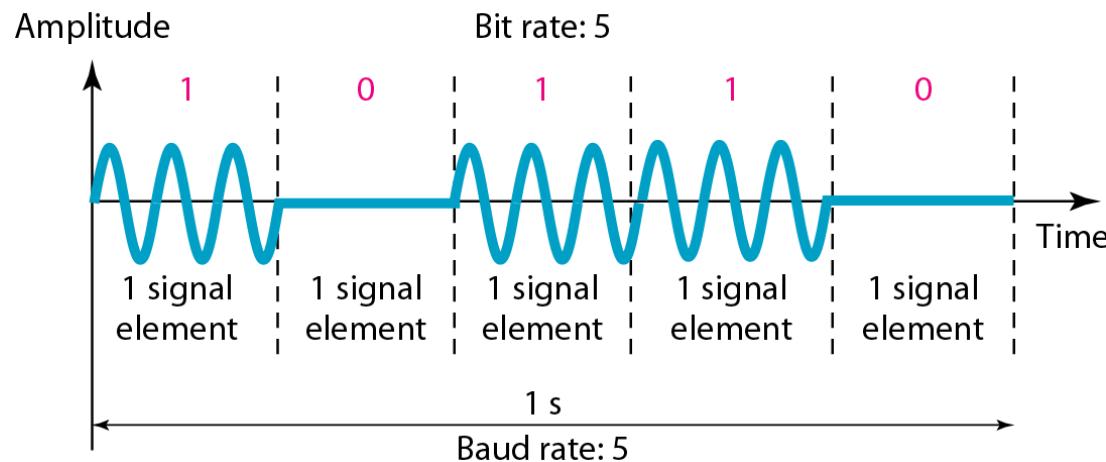


Fig. 3: Binary amplitude shift keying

Amplitude Shift Keying (ASK)

- ASK is implemented by **changing** the amplitude of a carrier signal to reflect amplitude levels in the digital signal.
- For example: a digital “**1**” could not affect the signal, whereas a digital “**0**” would, by making it zero.
- The **line encoding** will determine the values of the analog waveform to reflect the digital data being carried.
- The **bandwidth B** of ASK is proportional to the signal rate **S** .

$$B = (1+d)*S$$

- “ **d** ” is due to **modulation and filtering**, lies between **0 and 1** .

Implementation of Binary ASK

- Implementation of binary ASK:

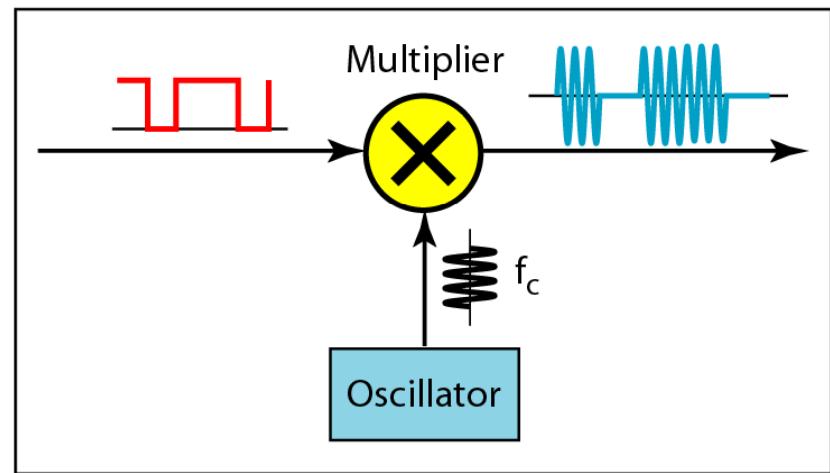
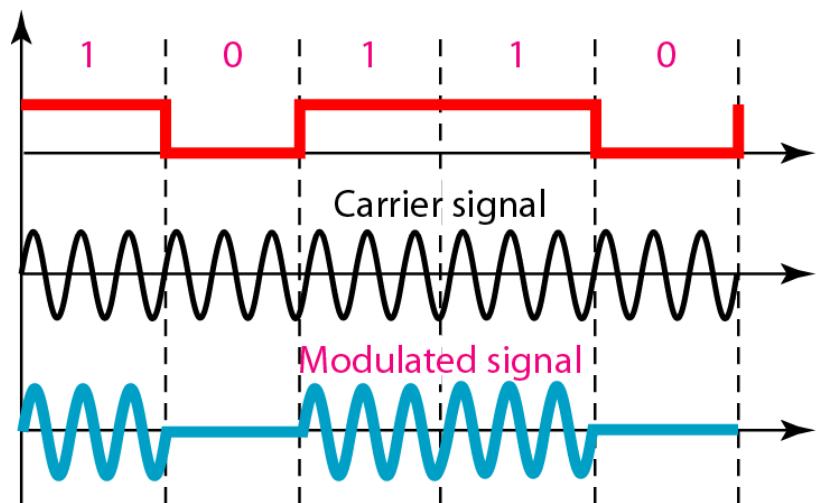


Fig. 4: Implementation of Binary ASK

ASK

- **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 – noise 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

ASK

- Example 3:

We have an available bandwidth of **100 kHz** which spans from **200 to 300 kHz**. What are the **carrier frequency** and the **bit rate** if we modulated our data by using **ASK** with **$d = 1$** ?

- Solution:

The middle of the bandwidth is located at **250 kHz**. This means that our carrier frequency can be at **$f_c = 250 \text{ kHz}$** . We can use the formula for bandwidth to find the bit rate (**with $d = 1$ and $r = 1$**).

$$B = (1 + d) \times S = 2 \times N \times \frac{1}{r} = 2 \times N = 100 \text{ kHz} \quad \rightarrow \quad N = 50 \text{ kbps}$$

ASK

- Example 4:

In data communications, we normally use **full-duplex** links with communication in both directions. We need to **divide** the bandwidth into two with **two carrier frequencies**, as shown in **Figure 5**. The figure shows the **positions** of **two carrier frequencies** and the **bandwidths**. The available bandwidth for each direction is now **50 kHz**, which **leaves** us with a data rate of **25 kbps** in each direction.

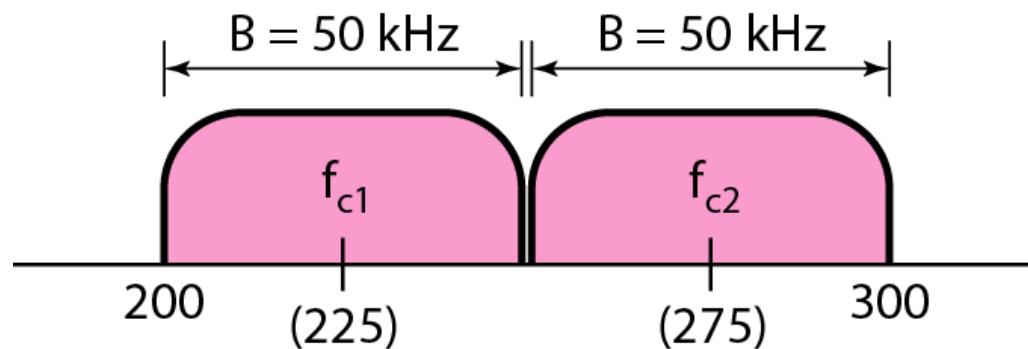


Fig. 5: Bandwidth of full-duplex ASK used in Example 4.

ASK

- **Multilevel ASK:**

The above discussion uses only **two amplitude levels**. We can have **multilevel ASK** in which there are more than two levels.

We can use **4, 8, 16, or more different amplitudes** for the signal and modulate the data using **2, 3, 4, or more bits at a time**.

In these cases, **r=2, r=3, r=4, and so on**. It is implemented with **QAM**.

Frequency Shift Keying (FSK)

- Frequency Shift Keying (FSK):

In Frequency Shift Keying (FSK), the **frequency** of the carrier signal is **varied** to represent digital data.

The **frequency** of the modulated signal is **constant** for the duration of **one signal element**, but **changes** for the next element if the **data element changes**.

Both **peak amplitude** and **phase** remain **constant** for all signal elements.

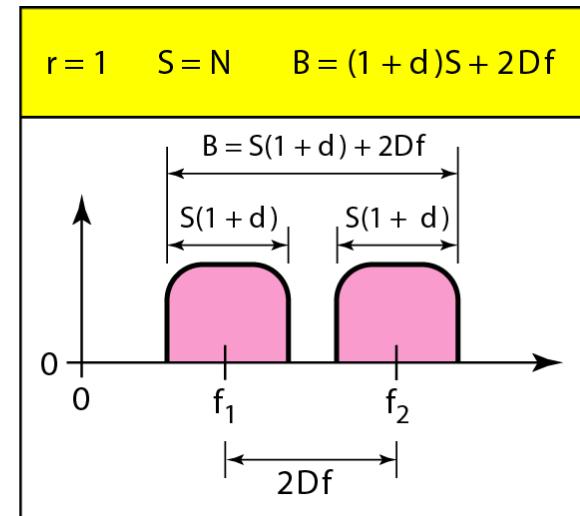
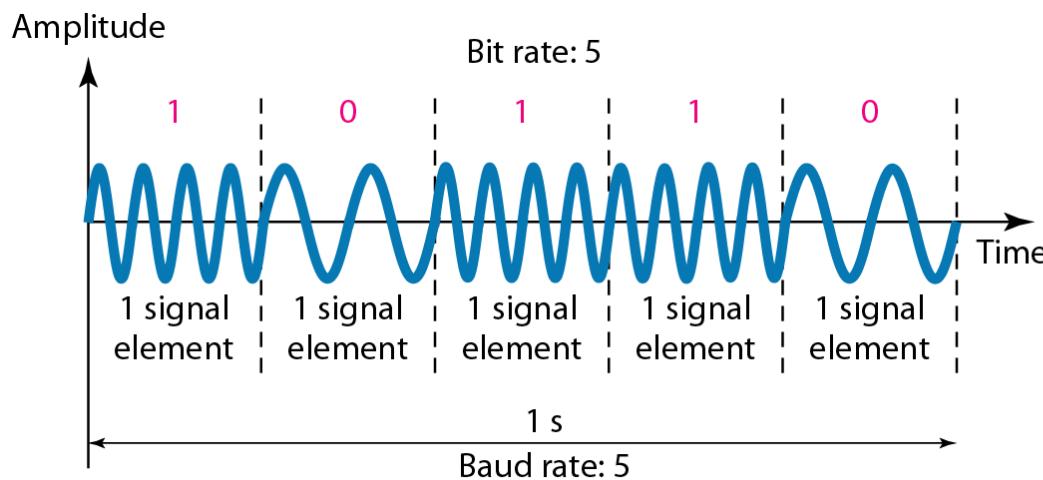


Fig. 6: Binary frequency shift keying

Frequency Shift Keying (FSK)

- The digital data stream changes the frequency of the carrier signal, f_c .
- For example, a “1” could be represented by $f_1=f_c + \Delta f$, and a “0” could be represented by $f_2=f_c - \Delta f$.
- If the difference between the two frequencies (f_1 and f_2) is $2\Delta f$, then the required $BW B$ will be:

$$B = (1+d) \times S + 2\Delta f$$

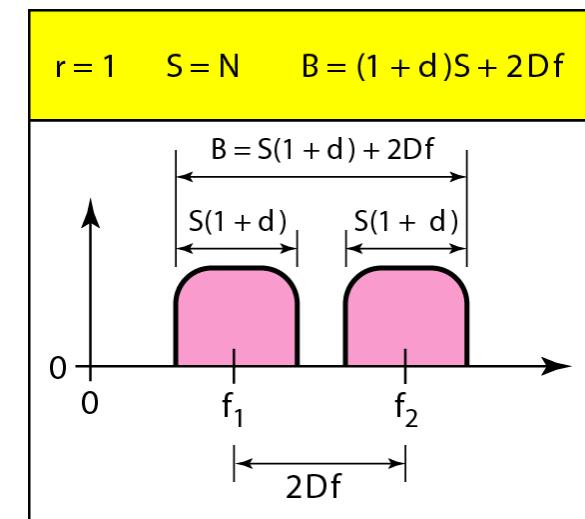
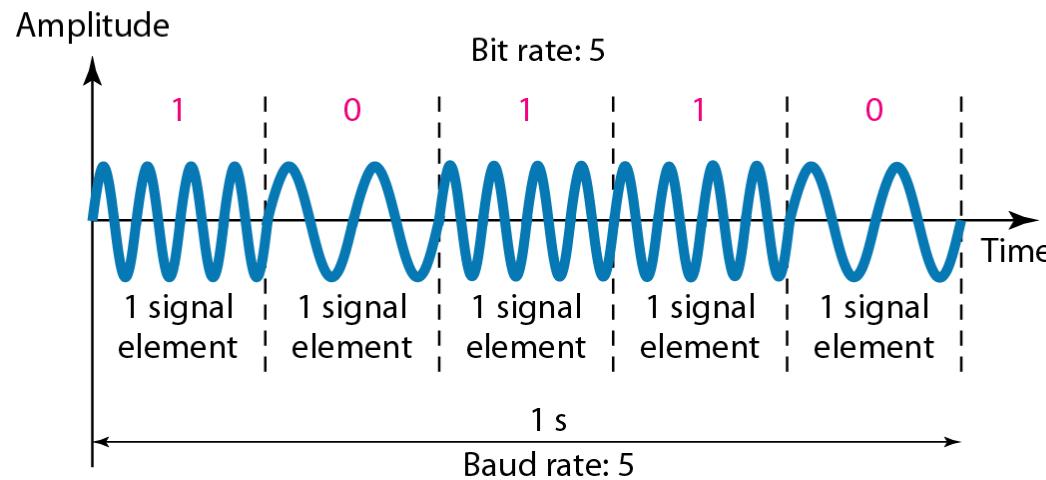


Fig. 6: Binary frequency shift keying

Implementation of binary FSK

- Implementation of binary FSK:

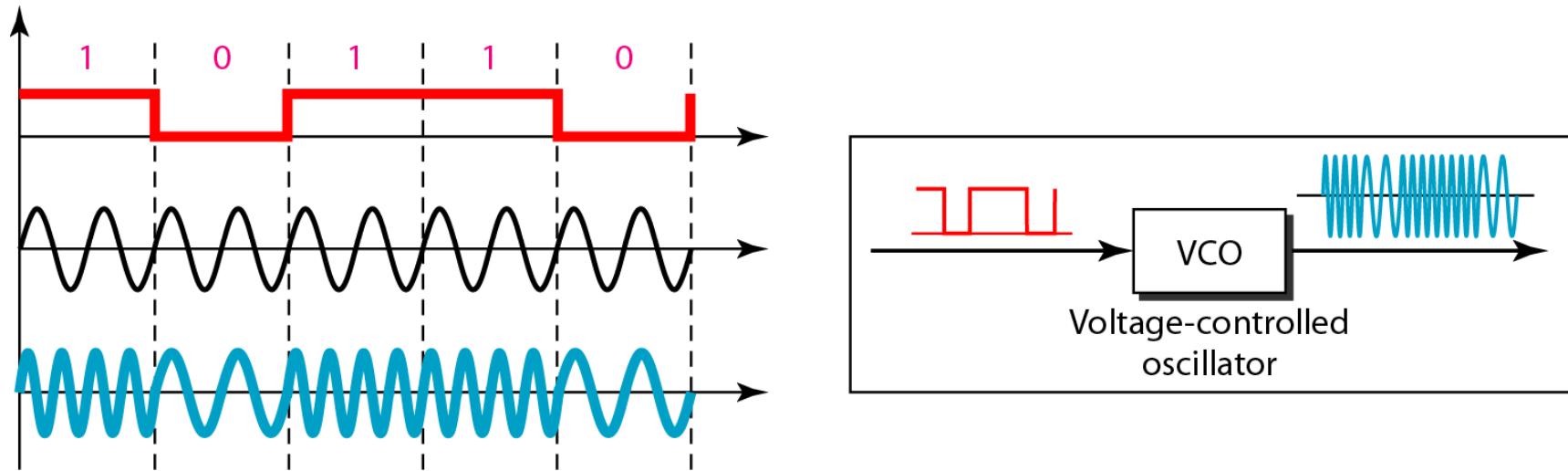


Fig. 7: Implementation of Binary FSK

Frequency Shift Keying (FSK)

- **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 looks 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-freq. radio transmission, etc.

Frequency Shift Keying (FSK)

- Example 5:

We have an available bandwidth of **100 kHz** which spans from **200 to 300 kHz**. What are the **carrier frequency** and the **bit rate** if we modulated our data by using **FSK** with **$d = 1$** ?

- Solution:

This problem is similar to **Example 3**, but we are modulating by using **FSK**. The **midpoint** of the band is at **250 kHz**. We choose **$2\Delta f$** to be **50 kHz**; this means

$$B = (1 + d) \times S + 2\Delta f = 100 \quad \rightarrow \quad 2S = 50 \text{ kHz} \quad S = 25 \text{ baud} \quad N = 25 \text{ kbps}$$

FSK

- **Multilevel FSK:**

In Multilevel FSK modulation (MFSK), we can use **more than two frequencies**.

For example, we can use **four** different frequencies **f_1, f_2, f_3 and f_4** to send **2 bits** at a time.

To send **3 bits** at a time, we can use **eight frequencies** and so on.

For the proper operation of the modulator and demodulator, it can be shown that the minimum value of **$2\Delta f$ need to be equal to S** . The bandwidth with **$d=0$** is,

$$B = (1+d) \times S + (L-1)2\Delta f = L \times S$$

FSK

- Example 6:

We need to send data **3 bits** at a time at a bit rate of **3 Mbps**. The carrier frequency is **10 MHz**. Calculate the **number of levels (different frequencies)**, the **baud rate**, and the **bandwidth**.

- Solution:

We can have $L = 2^3 = 8$. The baud rate is $S = 3 \text{ Mbps}/3 = 1 \text{ Mbaud}$. This means that the carrier frequencies must be **1 MHz apart ($2\Delta f = 1 \text{ MHz}$)**. The bandwidth is **$B = 8 \times 1 \text{ MHz} = 8 \text{ MHz}$** . **Figure 8** shows the allocation of frequencies and bandwidth.

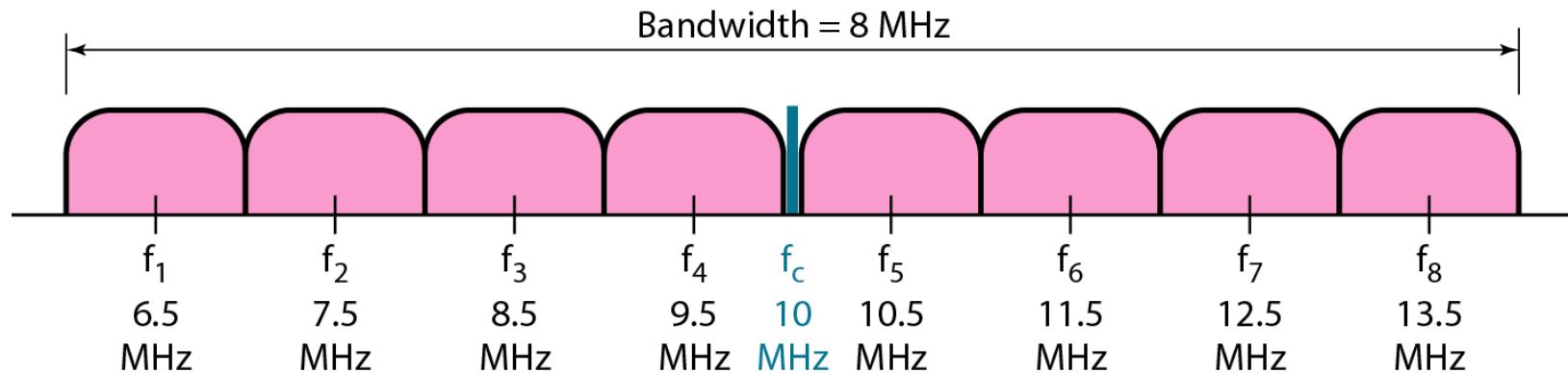


Fig. 8: Bandwidth of MFSK used in example 6.

Phase Shift Keying (PSK)

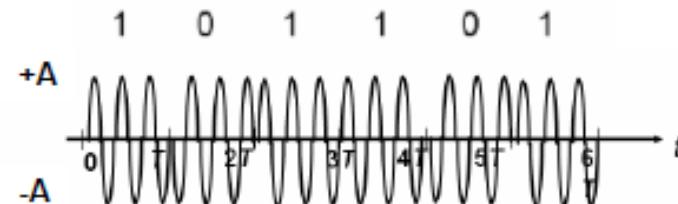
PSK – phase of carrier signal is varied to represent binary 1 or 0

- peak amplitude & freq. remain constant during each bit interval
- example: binary 1 = 0° phase, binary 0 = 180° (π rad) phase
⇒ PSK is equivalent to multiplying carrier signal by +1 when the information is 1, and by -1 when the information is 0

2-PSK, or
Binary PSK,
since only 2
different phases
are used.

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

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



- **demodulation:** demodulator must 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

Phase Shift Keying (PSK)

- Implementation of BPSK:

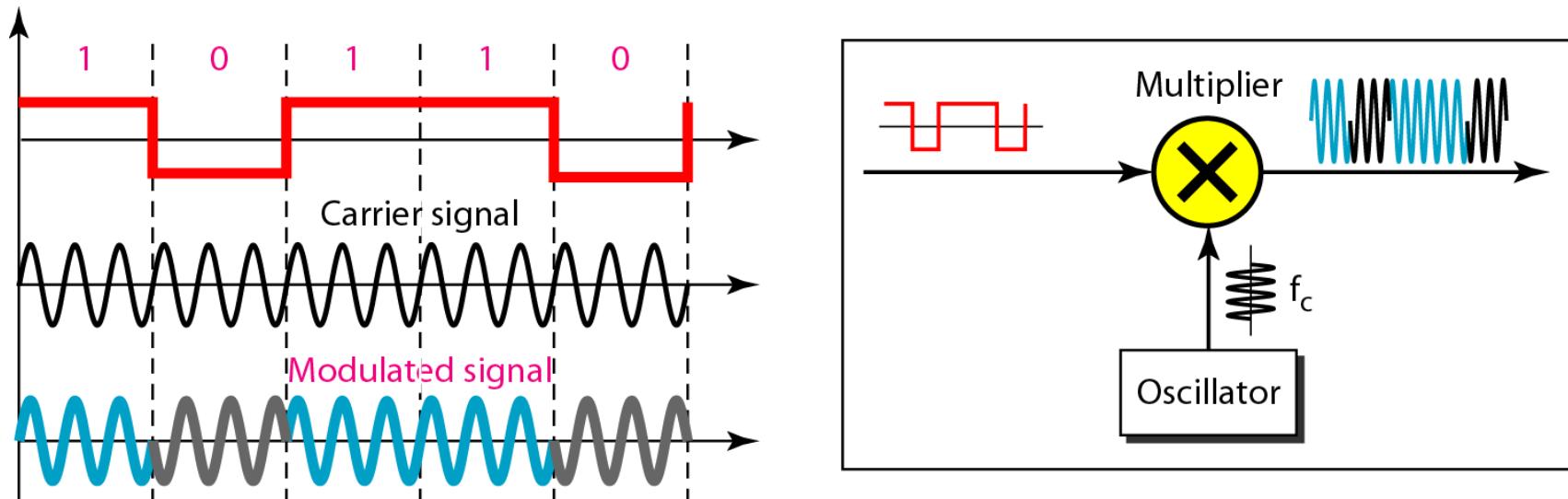
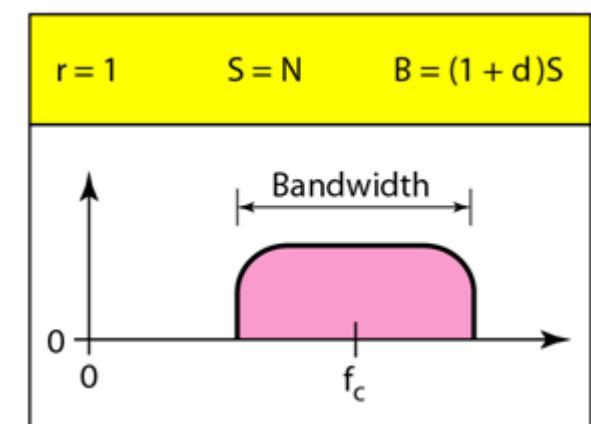


Fig. 10: Implementation of BPSK

The bandwidth requirement:

$$B = (1+d) \times S$$



Phase Shift Keying (PSK)

PSK Detection

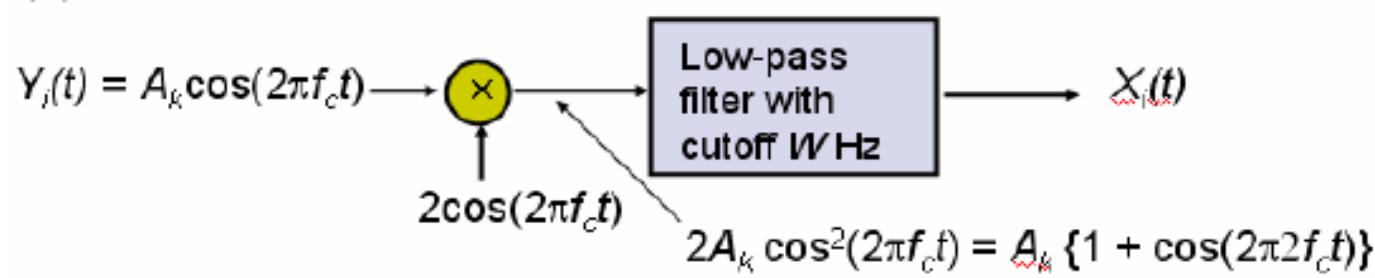
$$\cos^2 A = \frac{1}{2}(1 + \cos 2A)$$

- multiply the received / modulated signal $\pm A \cos(2\pi f_c t)$ by $2 \cos(2\pi f_c t)$
 - resulting signal

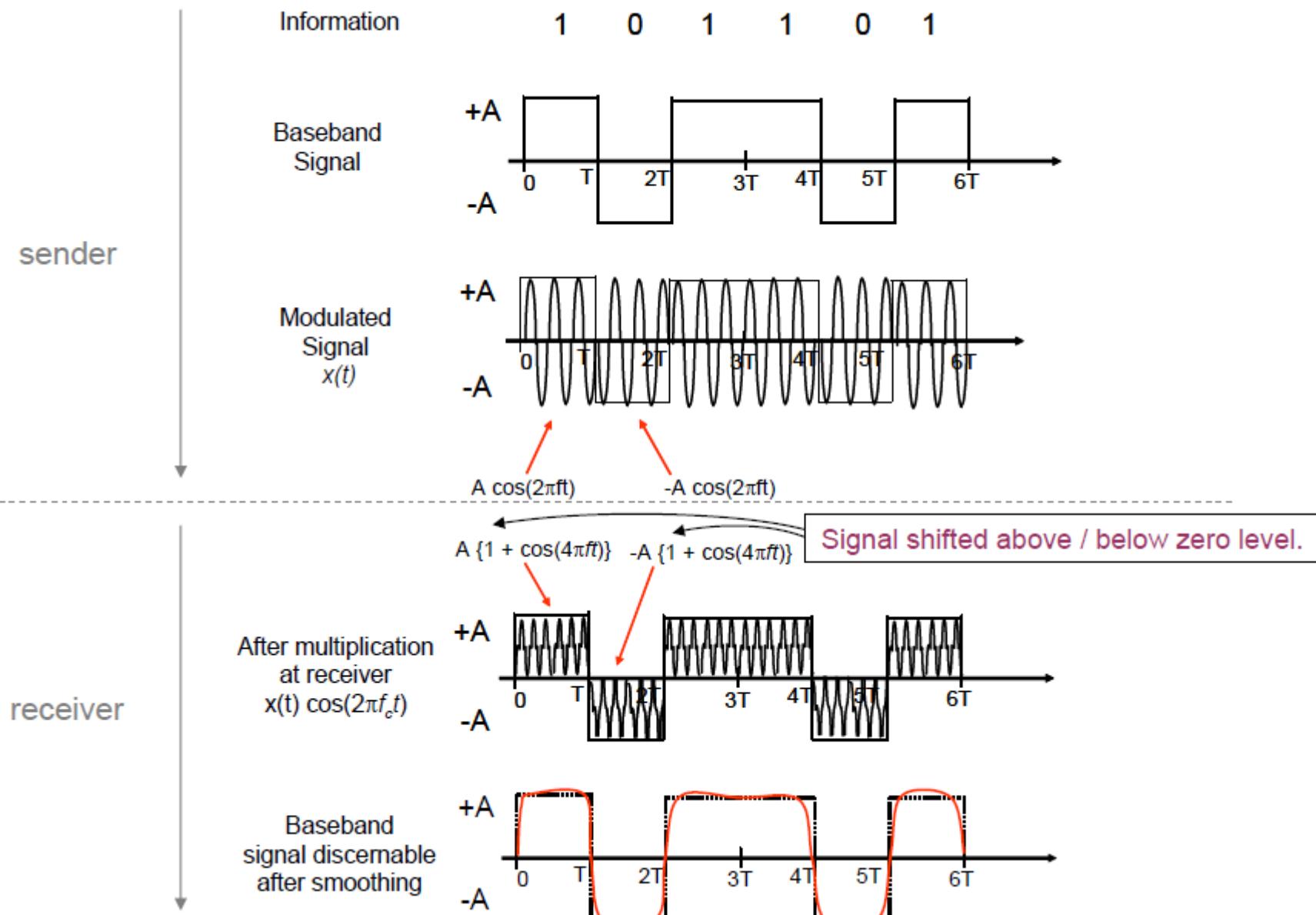
$$2A \cos^2(2\pi f_c t) = A[1 + \cos(4\pi f_c t)], \text{ binary 1}$$

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

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



Phase Shift Keying (PSK)



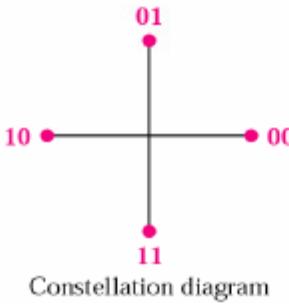
Quadrature PSK(QPSK)

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

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

Dabit	Phase
00	0
01	90
10	180
11	270

Dabit
(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

QPSK and its implementation

- QPSK and its implementation:

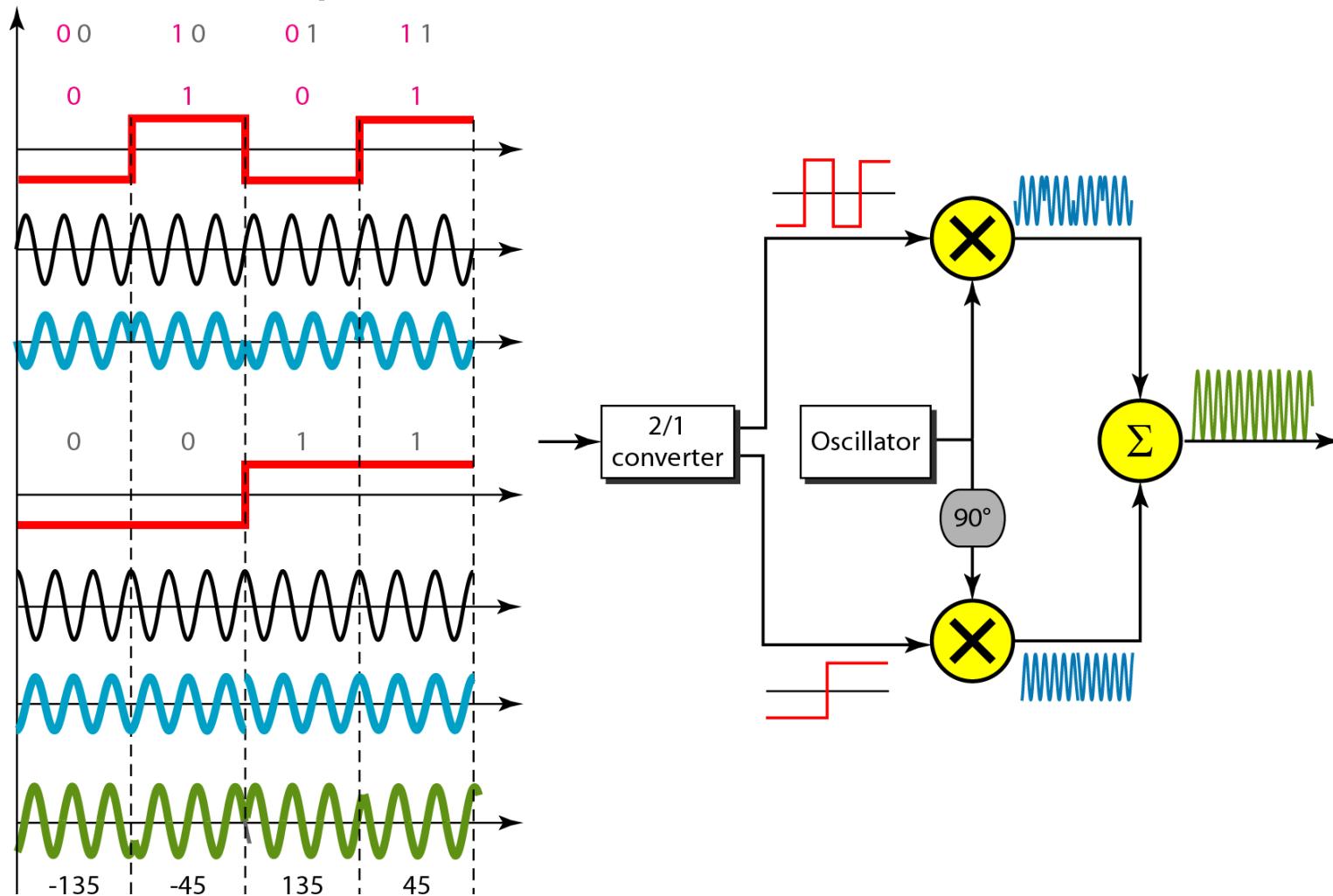


Fig. 11: Implementation of QPSK

Quadrature PSK(QPSK)

- Example 7:

Find the bandwidth for a signal transmitting at **12 Mbps** for **QPSK**. The value of **d = 0**.

- Solution:

For QPSK, **2 bits** is carried by one signal element. This means that **r = 2**. So the signal rate (baud rate) is **S = N × (1/r) = 6 Mbaud**. With a value of **d = 0**, we have **B = S = 6 MHz**.

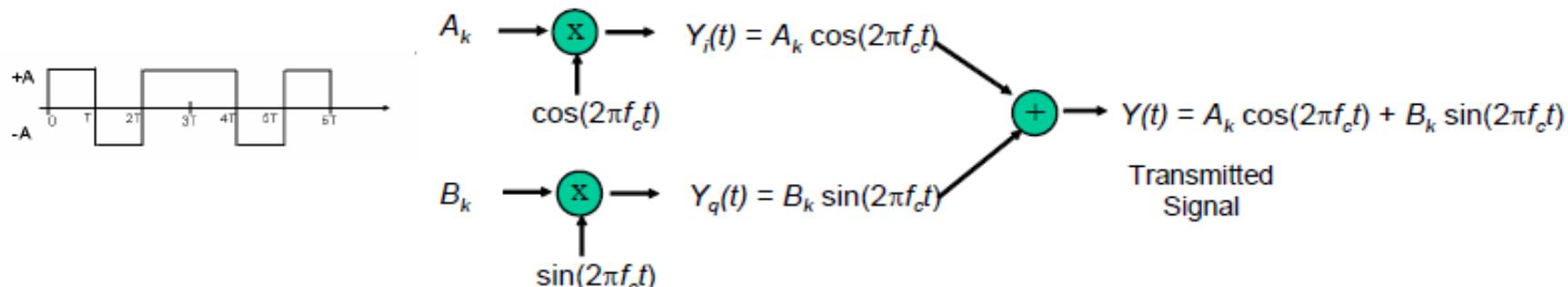
Quadrature Amplitude Modulation (QAM)

Quadrature Amplitude Modulation (QAM) – uses “two-dimensional” signalling

- original information stream is split into two sequences that consist of odd and even symbols, e.g. B_k and A_k

$$\begin{array}{ccccccc} 1 & 0 & 1 & 1 & 0 & 1 & \dots \\ 1 & -1 & 1 & 1 & -1 & 1 & \dots \\ B_1 & A_1 & B_2 & A_2 & B_3 & A_3 & \dots \end{array}$$

- A_k sequence (**in-phase comp.**) is modulated by $\cos(2\pi f_c t)$
- B_k sequence (**quadrature-phase comp.**) 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

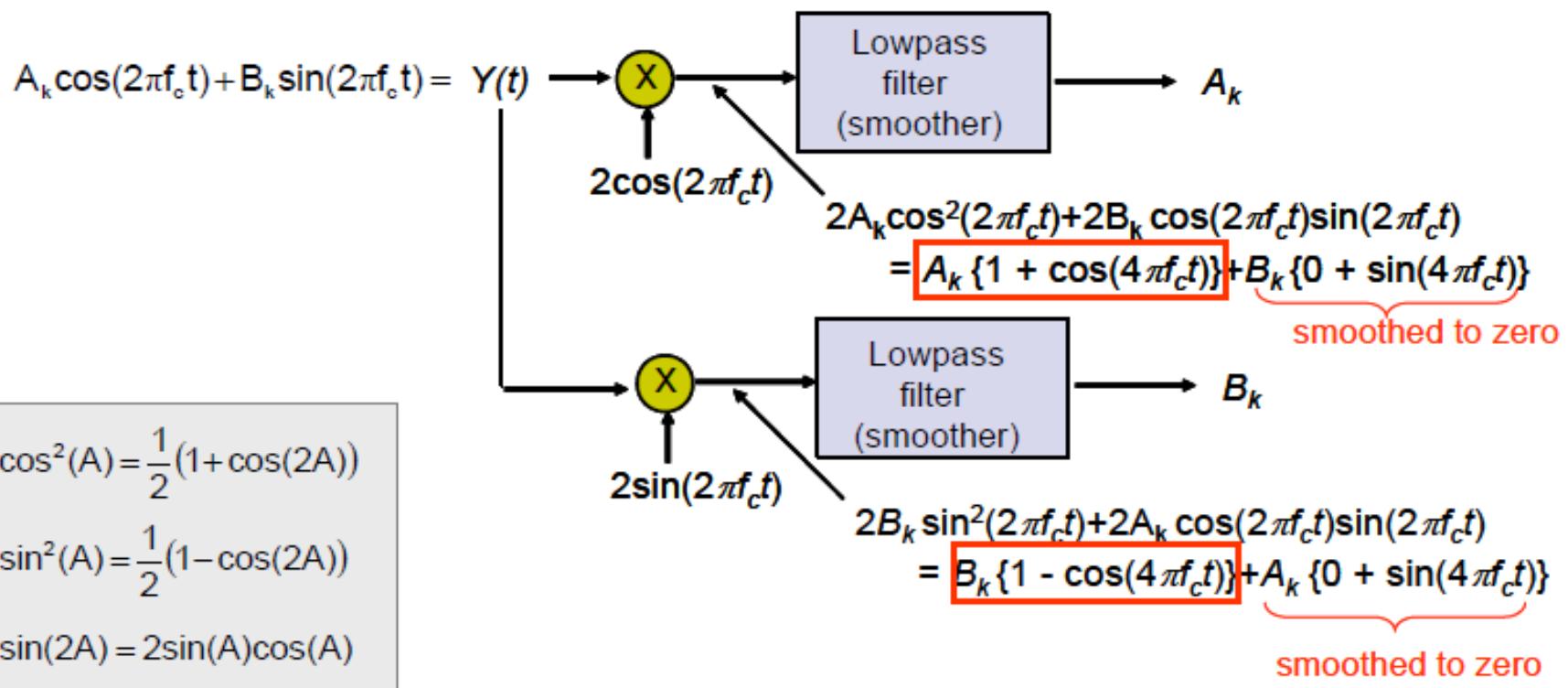


- advantage:** data rate = 2 bits per bit-interval!

Quadrature Amplitude Modulation (QAM)

QAM Demodulation

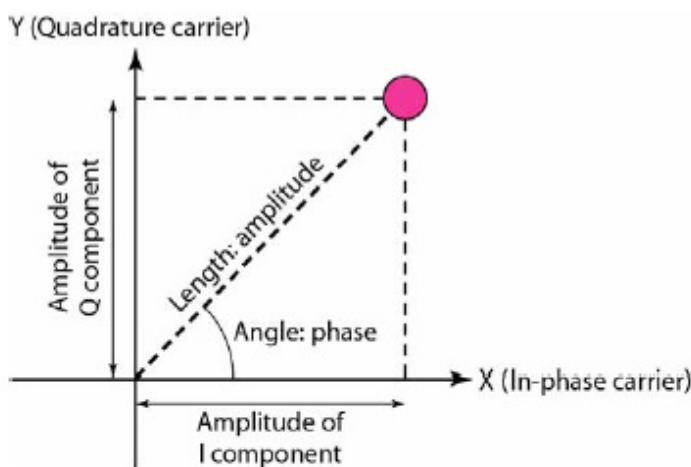
- by multiplying $Y(t)$ by $2 \cdot \cos(2\pi f_c t)$ and then low-pass filtering the resultant signal, sequence A_k is obtained
- by multiplying $Y(t)$ by $2 \cdot \sin(2\pi f_c t)$ and then low-pass filtering the resultant signal, sequence B_k is obtained



Signal Constellation

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 **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 line that connects the point to the origin is the peak amplitude of the signal element (combination of X & Y components)
- the angle the line makes with the X-axis is the phase of the signal element

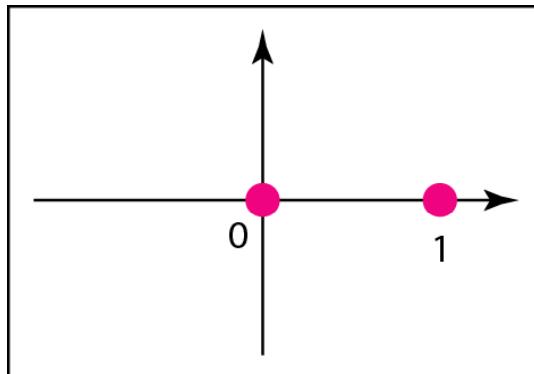


Constellation Diagrams

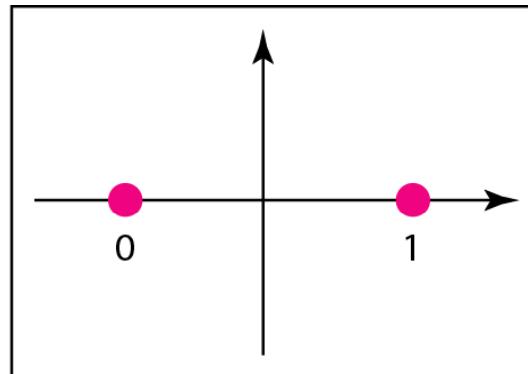
- Example 8:

Show the **constellation** diagrams for an **ASK (OOK)**, **BPSK**, and **QPSK** signals.

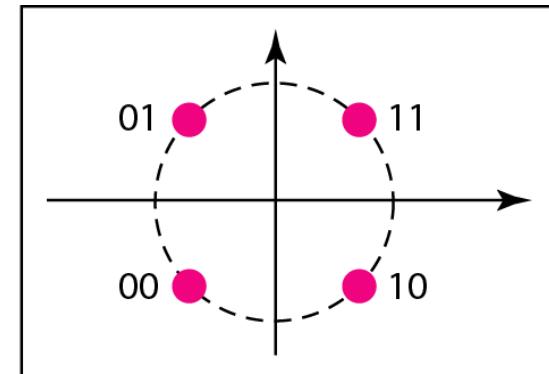
- Solution:



a. ASK (OOK)



b. BPSK

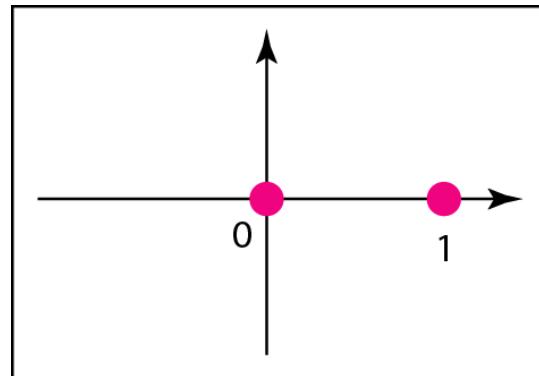


c. QPSK

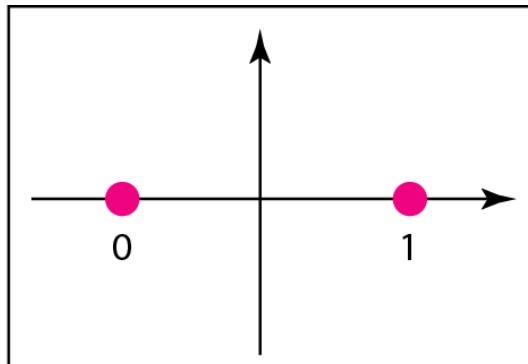
- For ASK, we are using only an in-phase carrier. Therefore, the **two points** should be on the **X axis**. **Binary 0** has an amplitude of **0V**; **binary 1** has an amplitude of **1 V**(for example). The points are located at the origin and at **1 unit**.

Constellation Diagrams

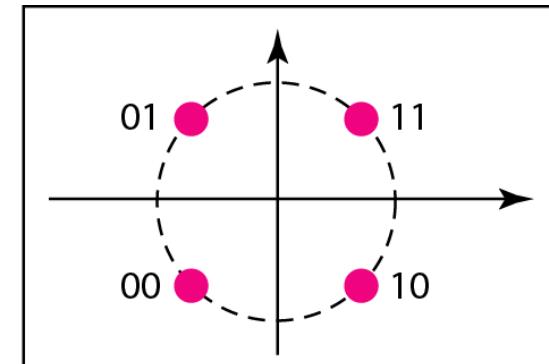
- **BPSK** creates two different signal elements, **one** with amplitude **1 V and in phase** and the **other** with amplitude **-1 V and 180° out of phase**.
- **QPSK** uses two carriers, one **in-phase** and the **other quadrature**. The point representing **11** is made of two combined signal elements, **both** with an amplitude of **1V**. **One element** is represented by an **in-phase carrier**, the other element by a **quadrature carrier**.



a. ASK (OOK)



b. BPSK



c. QPSK

Constellation Diagrams

- **Quadrature Amplitude Modulation (QAM):**

The **idea** of using two carriers, one in-phase and the other quadrature, **with different amplitude levels** for **each carrier** is the concept behind **quadrature amplitude modulation(QAM)**.

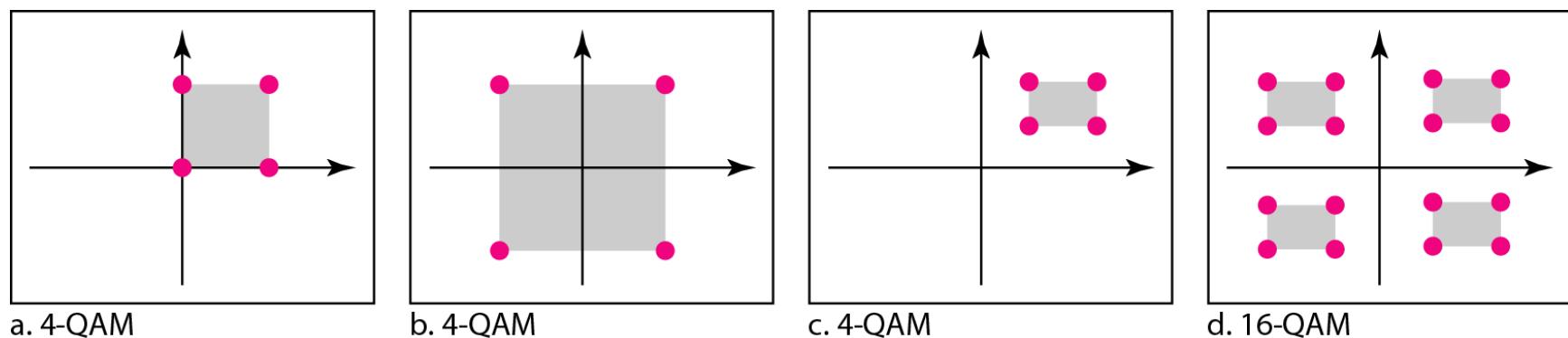


Fig. 13: Constellation diagrams for some QAMs

Continuous-phase frequency-shift keying (CPFSK)

- Continuous-phase frequency-shift keying (CPFSK) is a commonly used variation of frequency-shift keying (FSK), which is itself a special case of analog frequency modulation.
 - FSK is a method of modulating digital data onto a sinusoidal carrier wave, encoding the information present in the data to variations in the carrier's instantaneous frequency **between one of two frequencies** (referred to as the space frequency and mark frequency).
 - In general, a standard **FSK signal does not have continuous phase**, as the modulated waveform switches instantaneously between two sinusoids with different frequencies.
- As the name suggests, the phase of a CPFSK is in fact continuous;
 - this attribute is desirable for signals that are to be transmitted over **bandlimited channel**, as **discontinuities in a signal introduce wideband frequency components**.
 - In addition, some classes of amplifiers exhibit **nonlinear behavior** when driven **with nearly discontinuous signals**; this could have undesired effects on the shape of the transmitted signal.

Minimum Shift Keying (MSK)

- **Minimum Shift Keying (MSK):**

- Minimum shift keying (MSK) is a special type of continuous phase-frequency shift keying (CPFSK) with a modulation index of 0.5.
- A modulation index of 0.5 corresponds to the minimum frequency spacing that allows two FSK signals to be coherently orthogonal, and the name minimum shift keying implies the minimum frequency separation (i.e. bandwidth) that allows orthogonal detection.

- **Gaussian Minimum Shift keying (GMSK):**

- Gaussian Minimum Shift Keying (GMSK) is a modification of MSK (i.e. CPFSK with $m = 1/2$).
- A filter used to reduce the bandwidth of a baseband pulse train prior to modulation is called a pre-modulation filter.
- The Gaussian pre-modulation filter smoothens the phase trajectory of the MSK signal thus limiting the instantaneous frequency variations. The result is an **FM modulated signal** with a much **narrower bandwidth**. This bandwidth reduction does not come for free since the pre-modulation filter smears the individual pulses in pulse train. As a consequence of this smearing in time, adjacent pulses interfere with each other generating what is commonly called **inter-symbol interference or ISI**. In the applications where GMSK is used, the trade-off between power efficiency and bandwidth efficiency is well worth the cost.

Shannon's Theorem

- **Shannon's Theorem:**

Shannon's Theorem gives an upper bound to the capacity of a link, in bits per second (bps), as a function of the available bandwidth and the **signal-to-noise ratio** of the link.

The Theorem can be stated as:

$$C = B * \log_2(1 + S/N)$$

where **C** is the **achievable channel capacity**, **B** is the **bandwidth** of the line, **S** is the **average signal power** and **N** is the **average noise power**.

The **signal-to-noise ratio (S/N)** is usually expressed in **decibels (dB)** given by the formula:

$$10 * \log_{10}(S/N)$$

so for example a signal-to-noise ratio of **1000** is commonly expressed as

$$10 * \log_{10}(1000) = 30 \text{ dB}$$

END