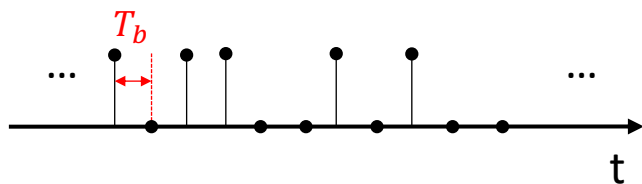


Digital communications

# How can we transmit a sequence of bits?

- What happens if we want to transmit a *sequence of bits* in the place of an analog signal?

$d_k = \dots 1, 0, 1, 1, 0, 0, 1, 0, 1, 0, 1, 0, 0, \dots$



$\sum_i d_k \delta(t - kT_b)$

- A train of delta occupies infinite bandwidth, before transmission the bits need to be passed through a low pass filter.
- Moreover, any signal transmitted in the air needs to be translated in frequency.

# How can we transmit a sequence of bits?

- Each bit of the sequence can be modelled as an equiprobable random variable

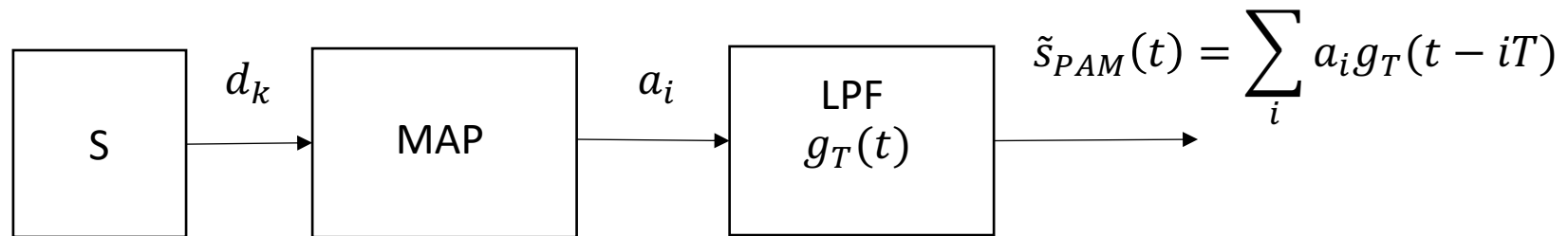
$$P\{d_k = 0\} = P\{d_k = 1\} = \frac{1}{2}$$

so that  $E\{d_k\} = \frac{1}{2}$ .

- In general, to save energy it is better to transmit 0-mean information. Bits  $d_k$  are mapped to 0-mean *information symbols*:  $a_i = 2d_i - 1$   
 $0 \rightarrow -1, \quad 1 \rightarrow 1$
- Using the same principles, one information symbol can be used to map more than just one bit.

# Pulse amplitude modulation

- Pulse amplitude modulation (PAM) is the modulation obtained by
  1. Mapping the bits  $d_k$  to the information symbols  $a_i$
  2. Filtering the symbols with a low pass filter with impulse response  $g_T(t)$



- Since the mapper can map a sequence of  $m$  bits on just one information symbol, the bit duration  $T_b$  and the symbol duration  $T$  may be different.

# Pulse amplitude modulation

- The signal  $\tilde{s}_{PAM}(t)$  is a *real* baseband signal that can be modulated at any frequency  $f_c$

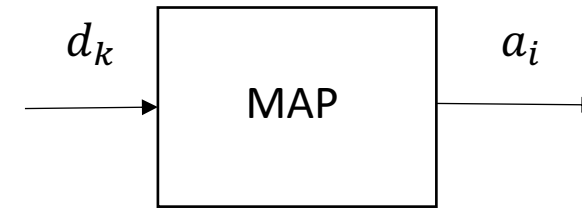
$$s_{PAM}(t) = \sum_i a_i g_T(t - iT) \cos(2\pi f_c t)$$

- The PAM signal is *equivalent* to an analog DSB where the modulating (and complex envelope) signal  $m(t)$  is

$$m(t) = \sum_i a_i g_T(t - iT) = \tilde{s}_{PAM}(t)$$

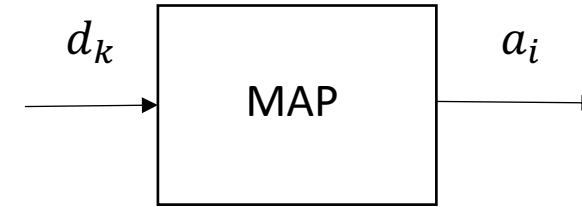
- From now on we will only consider the signal *baseband equivalent* and we will omit the tilde for ease of notation, i.e.  $s_{PAM}(t)$  indicates the complex envelope  $s_{PAM}(t) = \sum_i a_i g_T(t - iT)$ .

# PAM: symbol mapping



- The mapper block associates a sequence of  $m$  bits  $d_k$  to a single symbol  $a_i$ .
- Since each information symbol maps  $m$  bits, the symbol constellation contains  $M = 2^m$  symbols.  $M$  is always a power of 2.
- Usually, bit-to-symbol mapping is performed so that  $E\{a_i\} = 0$ .
- The larger the constellation size, the more bits are mapped to each symbol, the more efficient is the spectrum usage but also the larger is the average energy to transmit a bit.

# PAM: symbol mapping

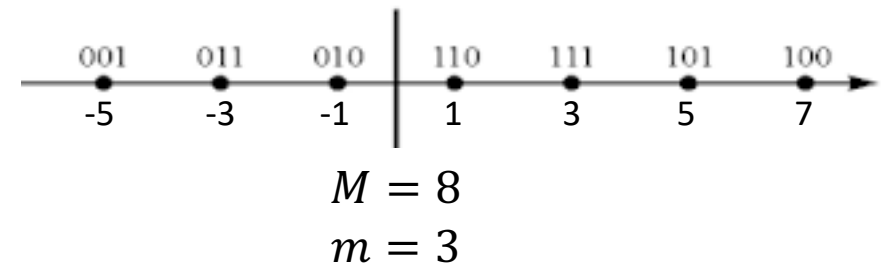
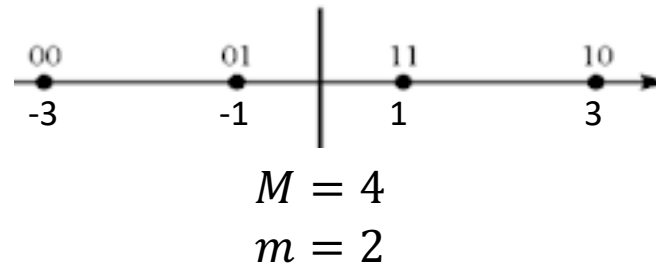
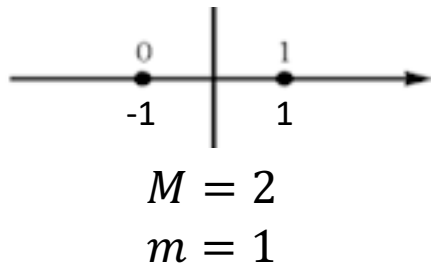


- The source generates bits with a rate  $R_b = \frac{1}{T_b}$ .
- Given the bit time  $T_b$ , since each symbol maps  $m$  bits, the symbol time  $T$  is

$$T = mT_b = \log_2 M T_b$$

- Accordingly, the symbol rate  $R$  is  $m$  times smaller than the bit rate  $R_b$

$$R = \frac{1}{T} = \frac{1}{mT_b} = \frac{R_b}{m} = \frac{R_b}{\log_2 M}$$



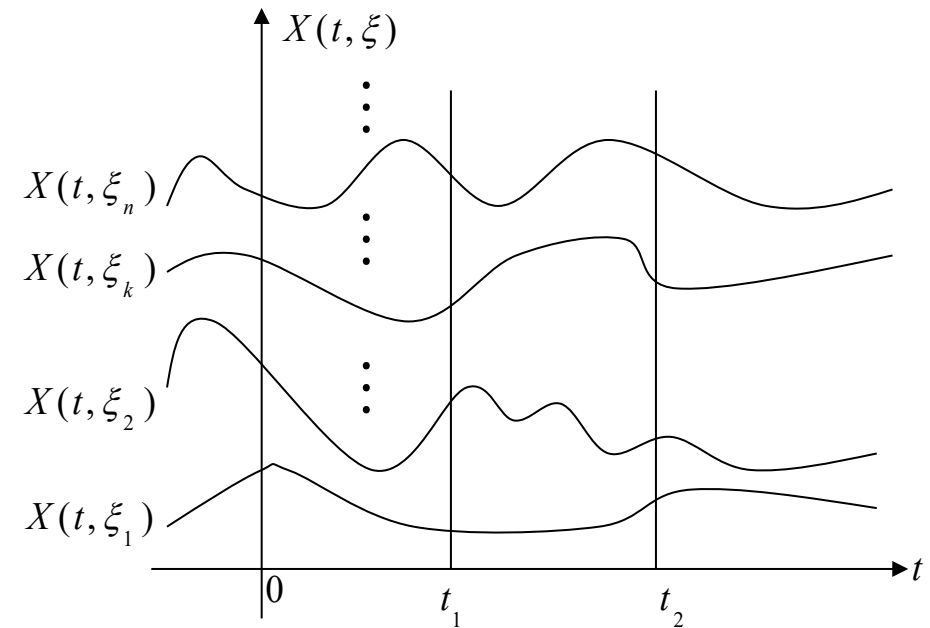
# Stochastic processes

- *Deterministic process.* A deterministic process is represented by an explicit mathematical relation.
- *Stochastic process.* A stochastic process is the result of a large number of separate causes, described in probabilistic terms and by properties which are averages.



# Stochastic processes

- A *stochastic process* is a set of random variables indexed in time
- Let  $\xi$  denote the random outcome of an experiment. To every such outcome suppose a waveform  $X(t, \xi)$  is assigned. The collection of such waveforms form a *stochastic process*.
- For a fixed  $\xi$  (the set of all experimental outcomes),  $X(t, \xi)$  is a specific time function.
- For fixed  $t = t_0$ ,  $X(t_0, \xi)$  is a random variable.
- The ensemble of all such realizations over time represents the stochastic process  $X(t)$ .



# Categories of stochastic processes

- *Parameter space*: set  $T$  of indices  $t \in T$ .
- *State space*: set  $S$  of values  $X(t) \in S$ .
- Categories:
  - Based on the parameter space:
    - Discrete-time processes: parameter space discrete,
    - Continuous-time processes: parameter space continuous.
  - Based on the state space:
    - Discrete-state processes: state space discrete,
    - Continuous-state processes: state space continuous.

# Distribution and probability density function

- If  $X(t)$  is a stochastic process, then for fixed  $t = t_0$ ,  $X(t_0)$  represents a *random variable*.

- The *distribution function* is given by

$$F_{X(t_0)}(x) = \Pr\{X(t_0) \leq x\}$$

$F_{X(t_0)}(x)$  depends on the value of the time instant  $t = t_0$ . For different values of  $t$ , we obtain a different random variable.

- Further, the first-order *probability density function* of the process  $X(t)$  is

$$f_{X(t_0)}(x) = \frac{d}{dx} F_{X(t_0)}(x)$$

# Independence

- For an *independent* stochastic process, the random variables obtained by sampling the process at any  $n$  times  $t_1, \dots, t_n$  are independent random variables for any  $n$ .

- Accordingly, the distribution is

$$\begin{aligned} F_{X(t_1), \dots, X(t_n)}(x_1, \dots, x_n) &= \Pr\{X(t_1) \leq x_1\} \cdots \Pr\{X(t_n) \leq x_n\} \\ &= F_{X(t_1)}(x_1) \cdots F_{X(t_n)}(x_n) \end{aligned}$$

and the probability density function is

$$f_{X(t_1), \dots, X(t_n)}(x_1, \dots, x_n) = f_{X(t_1)}(x_1) \cdots f_{X(t_n)}(x_n)$$

# Mean and autocorrelation

- **Mean** of a stochastic process:

$$\mu_X(t_0) = E\{X(t_0)\} = \int_{-\infty}^{+\infty} x f_{X(t_0)}(x) dx$$

is the mean value of the process  $X(t)$  at time  $t_0$ . In general, the mean of a process depends on the time index  $t$ .

- **Autocorrelation function** of a process:

$$R_{XX}(t_1, t_2) = E\{X(t_1)X^*(t_2)\} = \int_{-\infty}^{+\infty} \int_{-\infty}^{+\infty} x_1 x_2^* f_{X(t_1), X(t_2)}(x_1, x_2) dx_1 dx_2$$

represents the interrelationship between the random variables  $X_1 = X(t_1)$  and  $X_2 = X(t_2)$  obtained by sampling the process  $X(t)$  at times  $t_1$  and  $t_2$ .

# Stationarity

- A stationary process exhibits statistical properties that are invariant to shift in the time index.
- *First-order stationarity* implies that the statistical properties of  $X(t_0)$  and  $X(t_0 + c)$  are the same for any  $c$ .

$$f_{X(t_0)}(x) = f_X(x)$$

- The *mean* is a constant and does not depend on  $t_0$ .
- *Second-order stationarity* implies that the statistical properties of the pairs  $\{X(t_1), X(t_2)\}$  and  $\{X(t_1 + c), X(t_2 + c)\}$  are the same for any  $c$ .

$$f_{X(t_1), X(t_2)}(x_1, x_2) = f_X(x_1, x_2, t_2 - t_1)$$

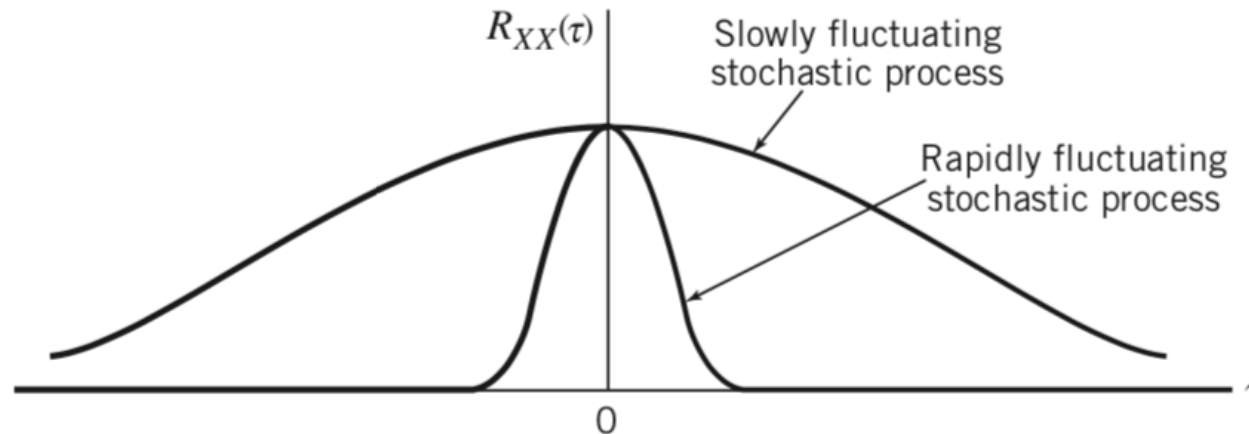
- The autocorrelation depends only on the difference of the time indices.

# Wide sense stationarity

- We can use a looser definition of stationarity. A process  $X(t)$  is said to be *wide-sense stationary* (WSS) if only these two conditions hold:
  - 1)  $E\{X(t)\} = \mu_X$
  - 2)  $E\{X(t_1), X(t_2)\} = R_{XX}(t_2 - t_1)$
- For a wide-sense stationary process, the mean is a constant and the autocorrelation function depends only on the difference between the time indices.

# Autocorrelation function

- The autocorrelation function  $R_{X,X}(t_1, t_2)$  describes the interrelation between two random variables obtained by sampling the stochastic process  $X(t)$  at times  $t_1$  and  $t_2$ .
- The more rapidly the stochastic process  $X(t)$  changes with time, the more rapidly will the autocorrelation function decrease from its maximum  $R_{XX}(0)$ .





# Power spectral density

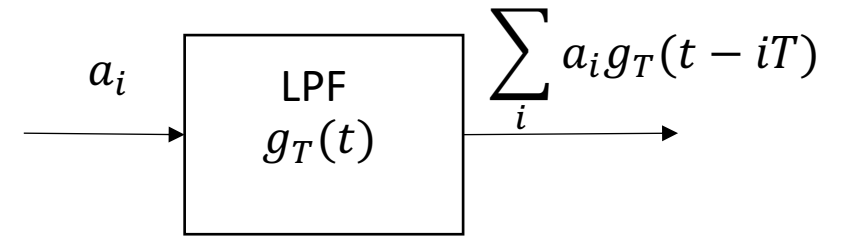
- The power spectral density  $S_{XX}(f)$  of a WSS stochastic process  $X(t)$  describes the distribution of power into frequency components composing that signal and is measured in watts per hertz (W/Hz).
- *Wiener-Kintchine theorem*. The power spectrum of  $X(t)$  is obtained as the Fourier transform of the autocorrelation function  $R_{XX}(\tau)$

$$S_{XX}(f) = \mathcal{F}\{R_{XX}(\tau)\} = \int_{-\infty}^{+\infty} R_{XX}(\tau) e^{-j2\pi f\tau} d\tau$$

- The signal power of  $X(t)$  can be computed as

$$P_X = \int_{-\infty}^{+\infty} S_{XX}(f) df$$

# PAM: power spectral density

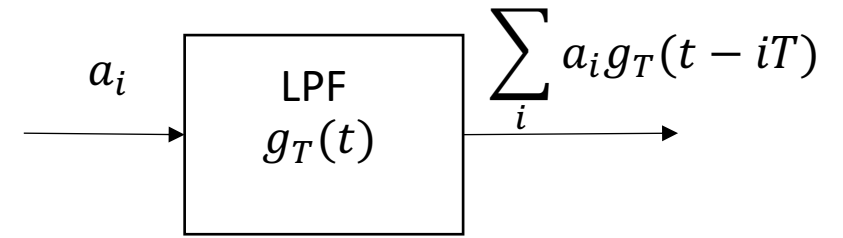


- A PAM signal is modelled as a *stochastic process* because the symbols  $a_i$  are samples of a discrete-time discrete-state stochastic process.
- The bandwidth occupied by a stochastic process is measured by its *power spectral density* (Fourier transform of its autocorrelation function).
- The PSD of the complex envelope  $s(t)$  of a PAM signal is

$$S_s(f) = \frac{1}{T} S_a(f) |G_T(f)|^2$$

where  $S_a(f)$  is the PSD of  $a_i$  and  $G_T(f)$  is the frequency response of the transmit filter  $g_T(t)$ .

# PAM: power spectral density



- $S_a(f)$  is computed as the Fourier transform of the autocorrelation function  $R_a(m)$  of the stationary, discrete, independent process  $a_i$ .

$$R_a(m) = E\{a_i a_{i+m}\} = \begin{cases} E\{a_i^2\} = A & m = 0 \\ (E\{a_i\})^2 & m \neq 0 \end{cases}$$

- When the symbols are zero-mean, i.e.,  $E\{a_i\} = 0$ , it is

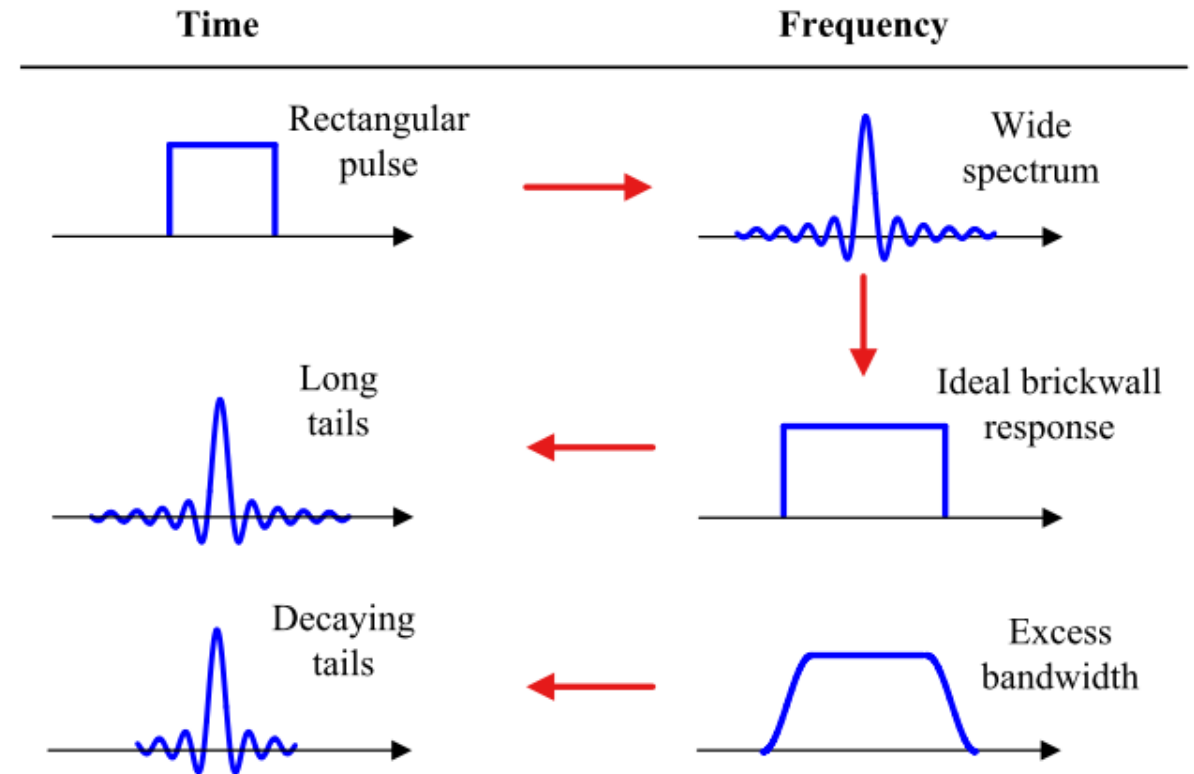
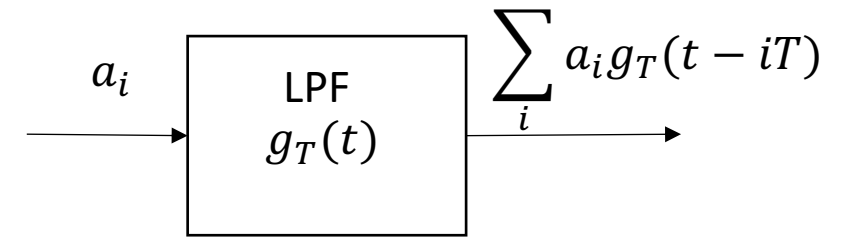
$$R_a(m) = A\delta(m) \Rightarrow S_a(f) = A,$$

and the PSD of the PAM signal is

$$S_s(f) = \frac{A}{T} |G_T(f)|^2$$

# PAM: pulse shaping

- The choice of the low-pass transmit filter determines the spectrum and the bandwidth of the PAM signal.
- If the pulse shape  $g_T(t)$  has a duration longer than 1 symbol time  $T$ , the spectrum is more compact but the energy of one symbols is spread over several intervals.



# PAM: pulse shaping

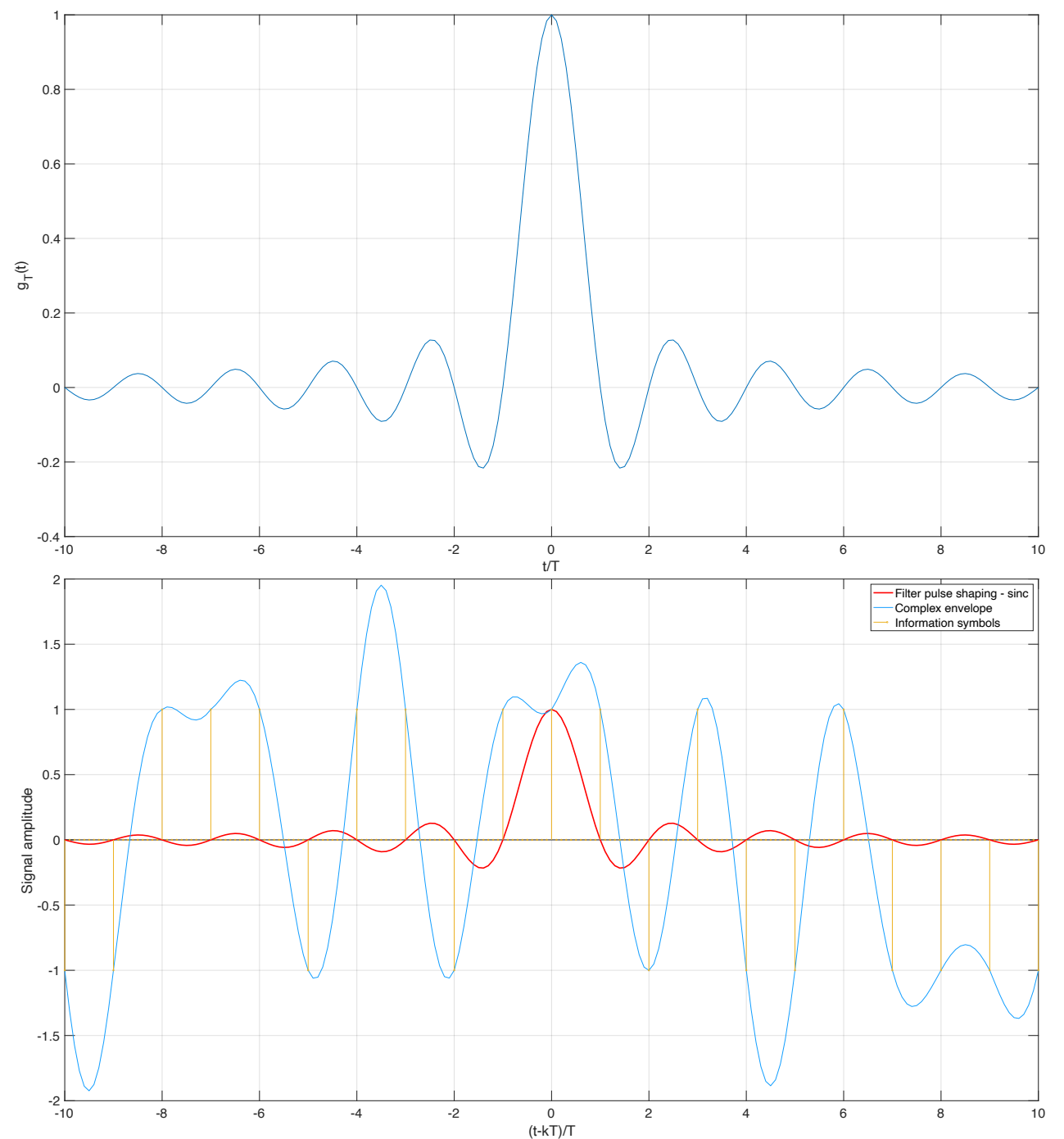
- The most compact spectrum is obtained when

$$G_T(f) = \text{rect}(fT),$$

which in the time domain corresponds to

$$g_T(t) = \frac{1}{T} \text{sinc}\left(\frac{t}{T}\right).$$

- The sinc spans an interval of several symbols.
- One single symbol 'mixes' its information with several adjacent symbols.
- This type of interference is denominated *inter-symbol interference* (ISI).

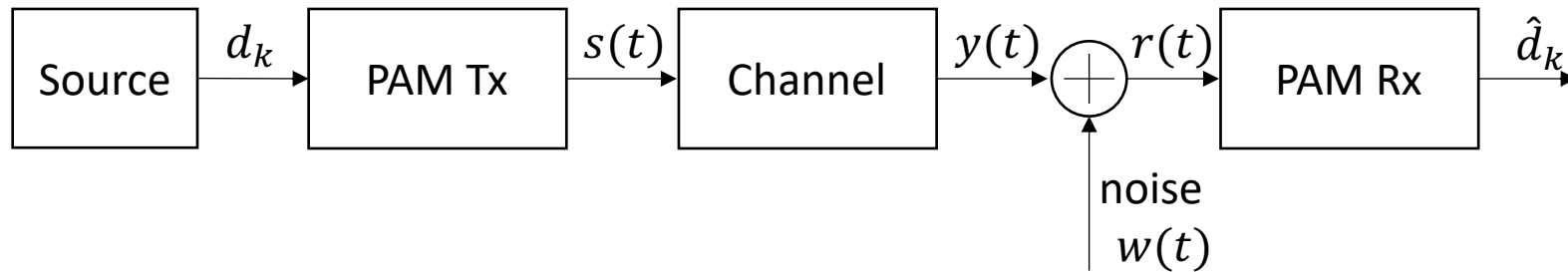


# PAM: occupied bandwidth

- There is a trade-off to make:
  - compact spectrum  $\rightarrow$  large amount of interference in the time domain (Extreme choice: a *rect* in the frequency domain and a *sinc* in time).
  - wide spectrum  $\rightarrow$  most of the symbol energy is contained within one symbol interval (Extreme choice: a *rect* in the time domain and a *sinc* in frequency).

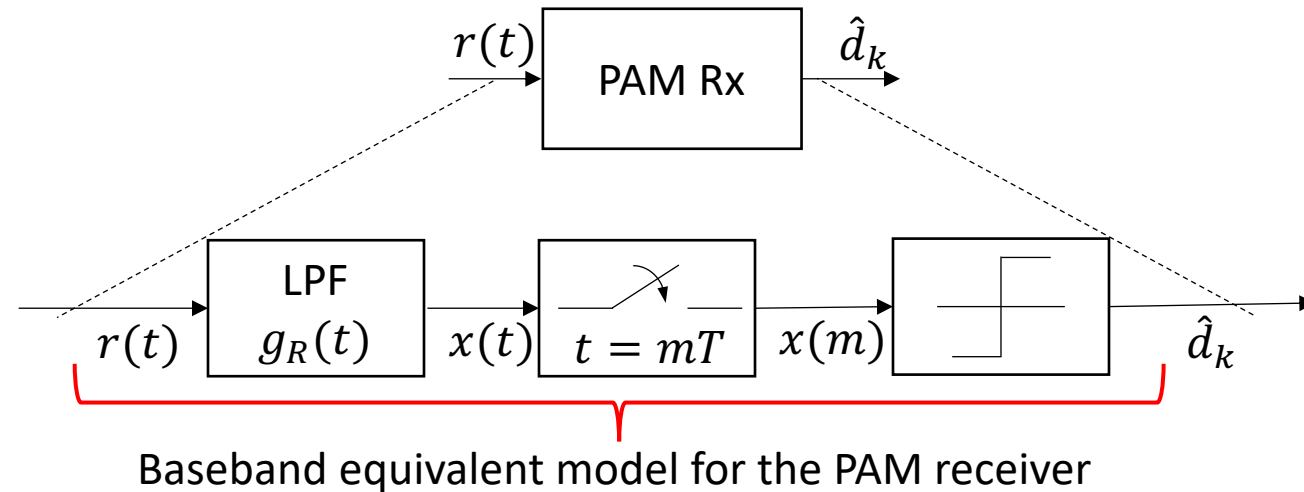
# PAM: receiver architecture

- PAM system block diagram



- The propagation channel is in general modelled as a LTI filter with impuls response  $h(t)$ . When the channel is ideal, it is  $h(t) = \delta(t)$ .
- The noise term is a white, zero-mean, Gaussian stationary process with PSD  $S_w(f) = N_0/2$  ( $S_w(f) = 2N_0$  for its complex envelope).
- The receiver's task is to reconstruct the sequence of transmitted bits from the received signal  $r(t)$ .

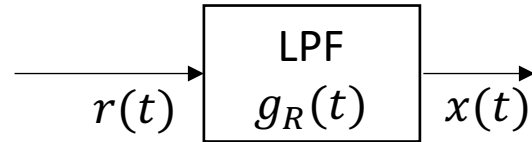
# PAM: receiver architecture



- The PAM receiver performs the inverse operation of the transmitter: extract the transmitted bits from the analog received signal  $r(t)$ .
  1. Filters noise and interference from the received signal  $r(t)$ ;
  2. Samples the filtered signal  $x(t)$  once per symbol time  $T$ ;
  3. Recovers the transmitted bits from the signal samples  $x(m)$ .



# PAM: Receive filter



- The received baseband equivalent signal has the form

$$r(t) = s(t) \otimes h(t) + w(t)$$

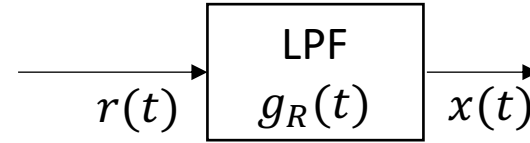
- The filter output is

$$x(t) = r(t) \otimes g_R(t) = \sum_i a_i g(t - iT) + n(t)$$

where

- $g(t) = g_T(t) \otimes h(t) \otimes g_R(t)$  is the convolution of the impulse response of the channel, the transmit and the receiver filter,
- $n(t)$  is the filtered (and colored!) noise.

# PAM receive filter



- One of the tasks of the receive filter  $g_R(t)$  is to remove the intersymbol interference affecting the received samples.
- The samples of the received signal take this form:

$$\begin{aligned} x(m) = x(t) \Big|_{t=mT} &= \sum_i a_i g(mT - iT) + n(mT) \\ &= a_m g(0) + \sum_{\ell, \ell \neq 0} a_{m-\ell} g(\ell T) + n(mT) \end{aligned} \quad \text{ISI}$$

- Neglecting the noise term, the condition on  $g(\ell T) = g(t)|_{t=\ell T}$  to have zero ISI is

$$g(\ell T) = \begin{cases} 1 & \ell = 0 \\ 0 & \ell \neq 0 \end{cases}$$

- Under these conditions (Nyquist criterion), the received sample  $x(m)$  is  
$$x(m) = a_m + n(mT)$$

# Nyquist criterion in the frequency domain

- The frequency response of the cascade of the channel, the transmit and the receive filter is  $G(f)$ , the Fourier transform of  $g(t)$ .
- Since sampling in time determines *periodicity* in the frequency domain,  $\mathcal{F}\{g(\ell)\}$ , the Fourier transform of  $g(\ell)$ ,  $g(t)$  sampled every  $T$  seconds, is

$$\mathcal{F}\{g(\ell)\} = \sum_{\ell} g(\ell) e^{-j2\pi f \ell T} = \frac{1}{T} \sum_k G\left(f - \frac{k}{T}\right)$$

# Nyquist criterion in the frequency domain

- On the other hand, if the sampled response  $g(\ell)$  satisfies the Nyquist criterion, then it is a Kronecker delta, i.e.  $g(\ell) = \delta(\ell)$ .
- The Fourier transform of  $\delta(\ell)$  is  $\mathcal{F}\{\delta(\ell)\} = 1$ .

- Accordingly, it is

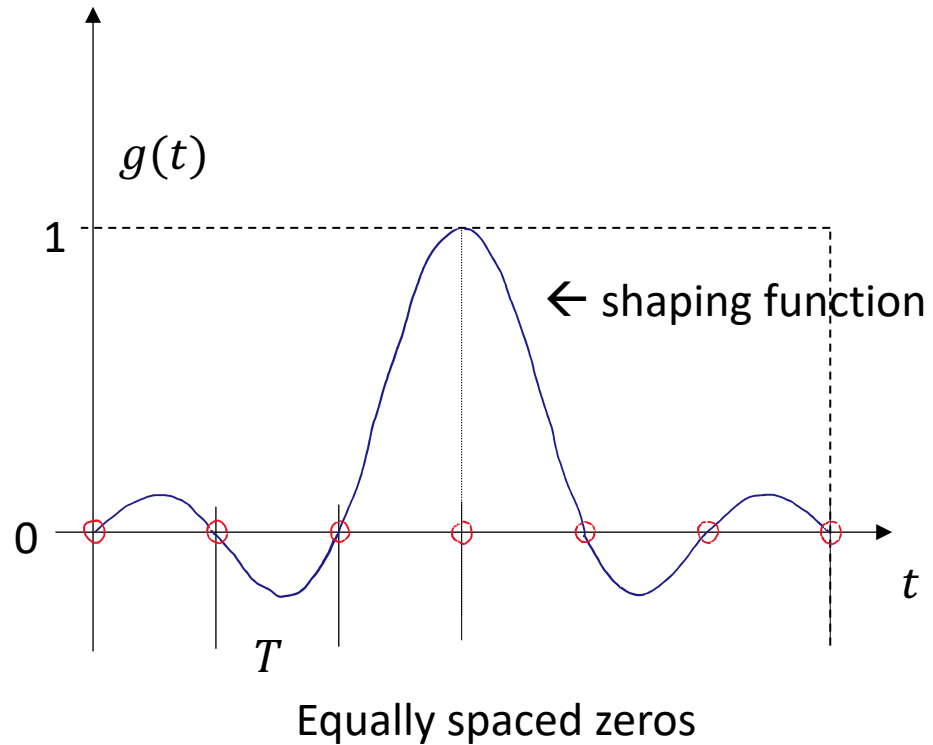
$$\mathcal{F}\{g(\ell)\} = \frac{1}{T} \sum_k G\left(f - \frac{k}{T}\right) = \mathcal{F}\{\delta(\ell)\} = 1$$

- From which we can extrapolate the Nyquist criterion for zero ISI in the frequency domain

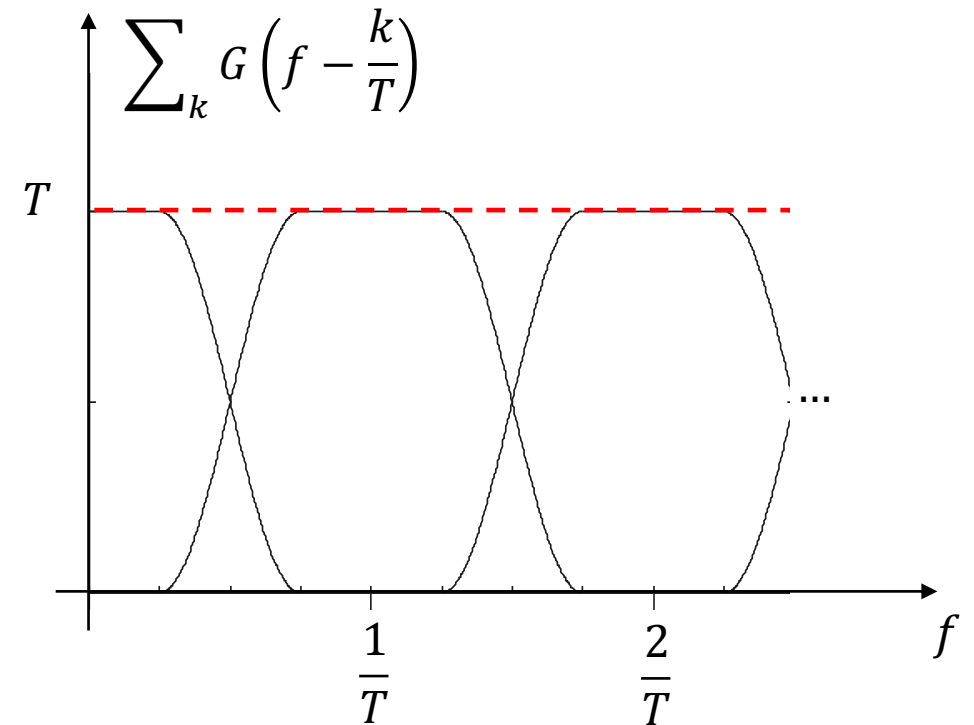
$$\sum_k G\left(f - \frac{k}{T}\right) = T$$

# Nyquist criterion

Time domain



Frequency domain

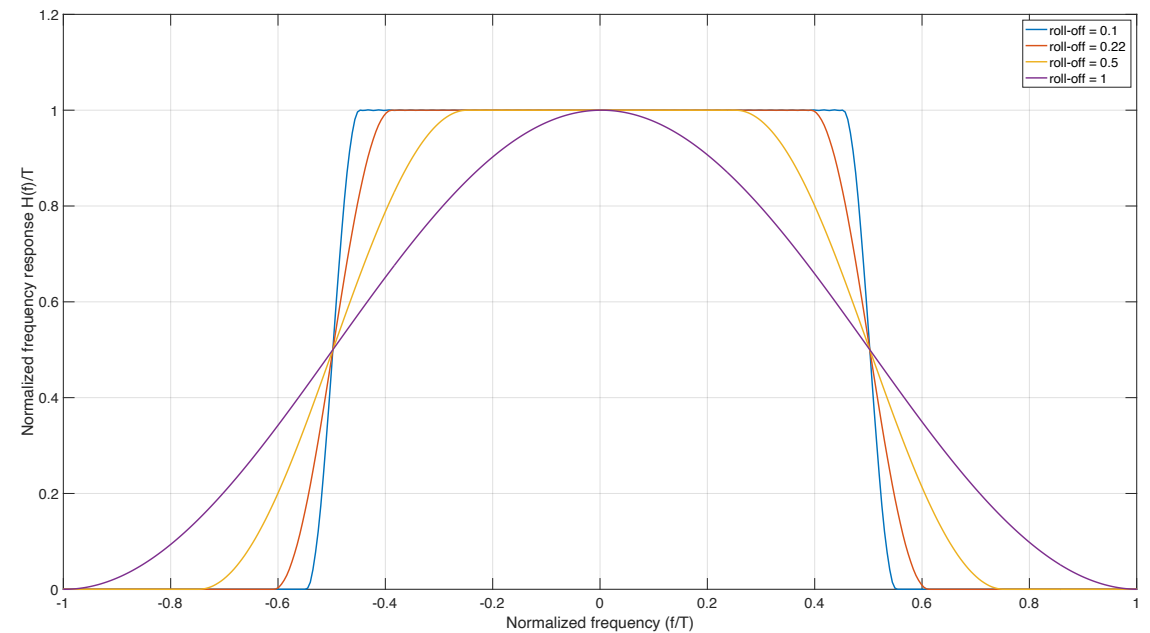
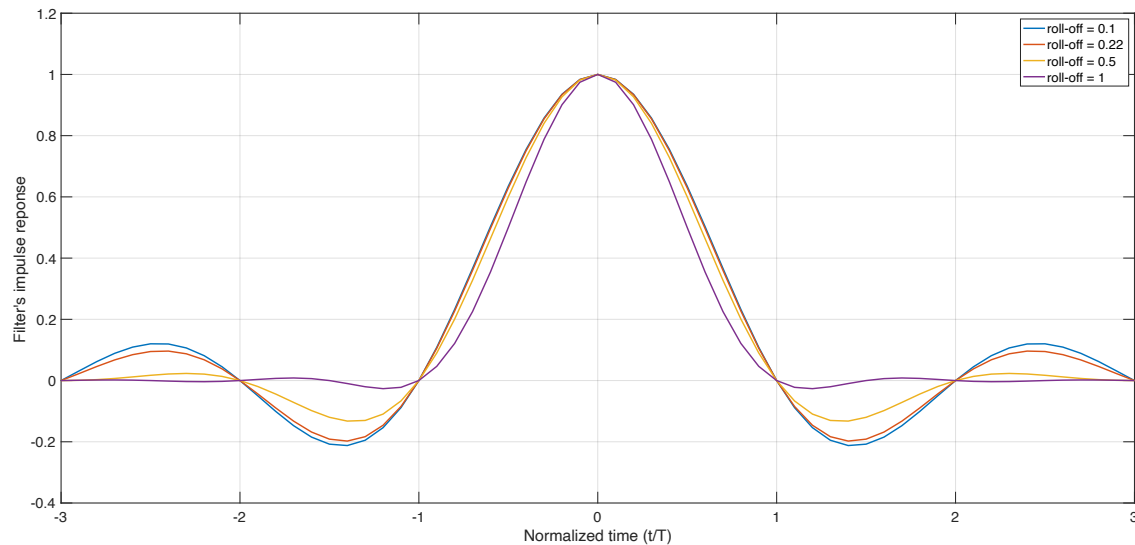


# Raised cosine filters

*Raised cosine* filters satisfy the Nyquist criterion: the occupied bandwidth is

$$B_{RC} = \frac{1 + \alpha}{2T}$$

The roll-off factor  $\alpha$  is a design parameter, RC with  $\alpha = 0$  is a rect and it is the *minimum bandwidth filter*.

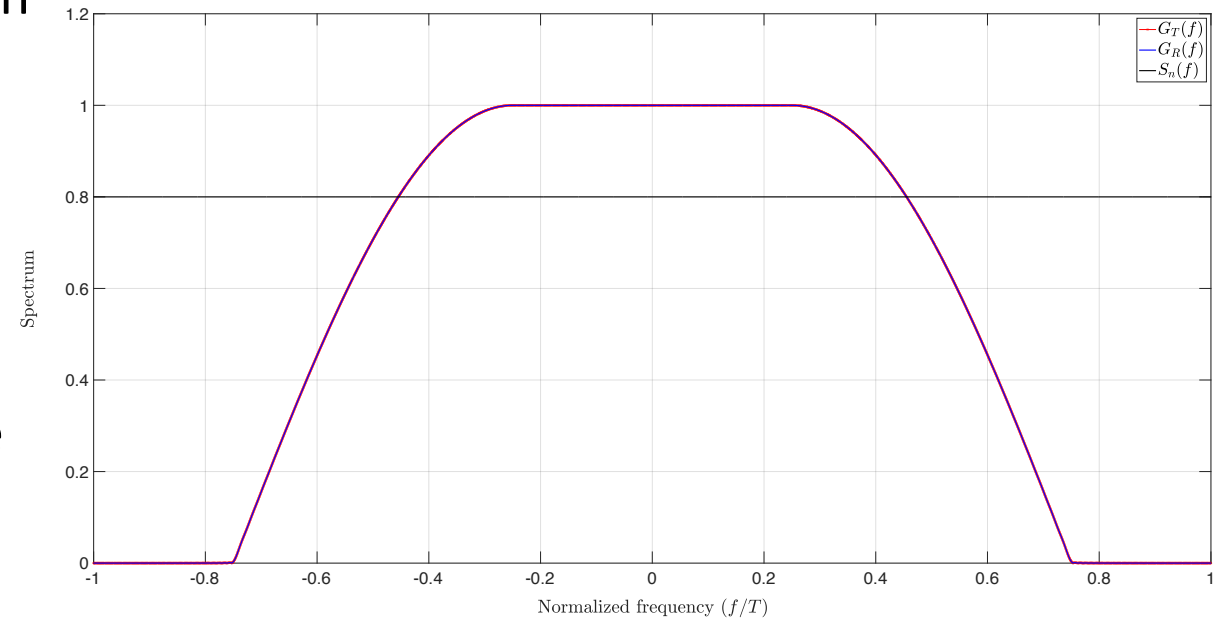


# Additive white Gaussian noise

- The complex envelope of the noise is  $w(t) = w_I(t) + jw_Q(t)$ , a stationary zero-mean white Gaussian process with PSD  $S_w(f) = 2N_0$  .
- The in-phase and quadrature components are independent and it is  $S_{w_I}(f) = S_{w_Q}(f) = N_0$ .
- The noise  $n(t) = n_I(t) + jn_Q(t) = g_R(t) \otimes w(t)$  is a zero-mean Gaussian complex stochastic process and its PSD is
$$S_n(f) = S_w(f)|G_R(f)|^2 = 2N_0|G_R(f)|^2$$
$$S_{n_I}(f) = S_{n_Q}(f) = N_0|G_R(f)|^2.$$

# Receive filter design: matched filter

- Neglecting, for the moment, the effect of the channel, the other major impairment at the receiver is the presence of Gaussian noise.
- The receiver should be designed to minimize the negative effect of Gaussian noise.
- The choice
$$g_R(t) = g_T(-t) \Leftrightarrow G_R(f) = G_T^*(f)$$
*maximizes* the signal-to-noise ratio at the receiver.
- The receiver filter is *matched* to the transmit filter.
- If  $g_T(t)$  is even, then it is  $g_R(t) = g_T(t)$ .





# Root raised cosine filters

- Root raised cosine filters are filters whose frequency response is the square root of a raised cosine, i.e.,  $H_{RRC}(f, \alpha) = \sqrt{H_{RC}(f, \alpha)}$ .
- If  $G_R(f) = G_T(f) = H_{RRC}(f, \alpha)$ , the transmit and receive filter pair satisfies the two independent *optimality* conditions:
  1. The cascade of  $g_T(t)$  and  $g_R(t)$  obeys the *Nyquist criterion*:
$$G_R(f)G_T(f) = \left(H_{RRC}(f, \alpha)\right)^2 = H_{RC}(f, \alpha).$$
  2. The receive filter is matched to the transmit filter.  
Since it is  $G_T(f) \in \Re \rightarrow G_R(f) = G_T(f) = G_T^*(f)$ , the transmit and receive filter are *matched*.

# Bandwidth of a PAM signal with RRC filtering

- If  $G_T(f) = H_{RRC}(f, \alpha)$  and the symbols are zero-mean, it is

$$S_s(f) = \frac{1}{T} S_a(f) |G_T(f)|^2 = \frac{A}{T} H_{RC}(f, \alpha).$$

- The bandwidth occupied by the PAM *complex envelope* is

$$B_{PAM}^{(BB)} = \frac{1 + \alpha}{2T} = \frac{1 + \alpha}{2} \frac{1}{\log_2 M T_b} = \frac{1 + \alpha}{2} \frac{R_b}{\log_2 M}$$

- The bandwidth occupied by the corresponding *passband signal* is

$$B_{PAM}^{(PB)} = 2B_{PAM}^{(BB)} = \frac{1 + \alpha}{T} = (1 + \alpha) \frac{R_b}{\log_2 M}$$

# Power of a PAM signal with RRC filtering

- The mean power of the *complex envelope* of a PAM signal with zero-mean symbols and root raised cosine filtering is

$$P_s^{(BB)} = \int_{-\infty}^{+\infty} S_s(f) df = \frac{A}{T} \int_{-\infty}^{+\infty} |G_T(f)|^2 df = \frac{A}{T} \int_{-\infty}^{+\infty} H_{RC}(f, \alpha) df$$

since it is  $\int_{-\infty}^{+\infty} H_{RC}(f, \alpha) df = h_{RC}(t, \alpha)|_{t=0} = 1$ ,

$$P_s^{(BB)} = \frac{A}{T}$$

- The power of the corresponding *passband signal* is

$$P_s = \frac{1}{2} P_s^{(BB)} = \frac{A}{2T}$$

# Energy of a PAM symbol with RRC filtering

- The mean square value of the symbols for a PAM constellation is

$$A = E\{a_i^2\} = \frac{M^2 - 1}{3}$$

- The energy per symbol is computed as the power multiplied by the symbol duration

$$E_s = P_s T = \frac{A}{2T} T = \frac{M^2 - 1}{6}$$

# Additive white Gaussian noise

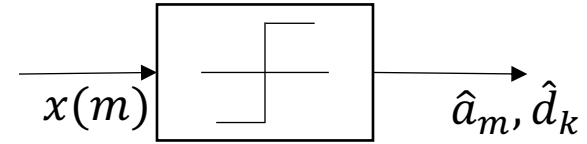
- The sample  $n(m) = n(t)|_{t=mT} = n_I(m) + jn_Q(m)$  is a zero-mean Gaussian complex *random variable* and its variance is

$$\sigma_n^2 = E\{|n(m)|^2\} = \int_{-\infty}^{+\infty} S_n(f) df = 2N_0 \int_{-\infty}^{+\infty} |G_R(f)|^2 df$$

- If the receive filter is a RRC, then it is  $\int_{-\infty}^{+\infty} |G_R(f)|^2 df = 1$  and
$$\sigma_n^2 = 2N_0$$

- The in-phase and quadrature components  $n_I(m)$  and  $n_Q(m)$  are independent and the variance of each component is  $\sigma^2 = \sigma_{n_I}^2 = \sigma_{n_Q}^2 = N_0$ .

# Decision strategy



- Under the hypothesis of RRC filtering at the transmit and at the receiver, the decision variable is

$$x(m) = a_m + n(m)$$

- The optimal decision strategy is the one that chooses the symbol that maximizes the probability conditioned on having received  $x(m)$ .

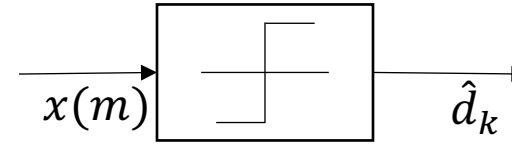
$$\hat{a}_m = \arg \max_{a^{(i)} \in \mathcal{A}} p(a^{(i)} | x(m))$$

- It can be shown that in case of equiprobable symbols it is

$$p(a^{(i)} | x(m)) \approx p(x(m) | a^{(i)})$$

so that the symbol  $a^{(i)}$  that maximizes  $p(x(m) | a^{(i)})$  maximizes also  $p(a^{(i)} | x(m))$ , *maximum likelihood decision*.

# PAM decision strategy



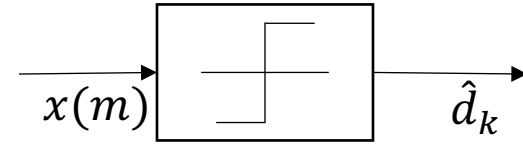
- Since PAM signal is *real*, we consider only the in-phase component of the received signal, i.e.  $x(m) = x_I(m)$  and  $n(m) = n_I(m)$ .
- Because of the conditioning, the symbol value  $a^{(i)}$  is fixed and  $x(m) \in \mathcal{N}(a^{(i)}, N_0)$  is a Gaussian random variable with mean  $a^{(i)}$  and variance  $\sigma^2 = \sigma_{n_I}^2 = N_0$ .
- The probability density function is

$$p(x(m)|a^{(i)}) = \frac{1}{\sqrt{2\pi}\sigma} e^{-\frac{(x(m)-a^{(i)})^2}{2\sigma^2}}$$

so that the symbol  $a^{(i)}$  that maximizes  $p(x(m)|a^{(i)})$  is the one that minimizes the distance between the symbol and the received sample

$$\hat{a}_m = \arg \min_{a^{(i)} \in \mathcal{A}} |x(m) - a^{(i)}|$$

# Decision strategy



- Example. Consider a 2-PAM ( $M = 2$ ) and assume that  $x(m) = 0.5$ . The two symbols are  $a^{(0)} = -1$  and  $a^{(1)} = 1$ , i.e.  $\mathcal{A} = \{-1, 1\}$ .
- The decision block will compute the two distances:

$$d(x(m), a^{(0)}) = |0.5 - (-1)| = 1.5$$

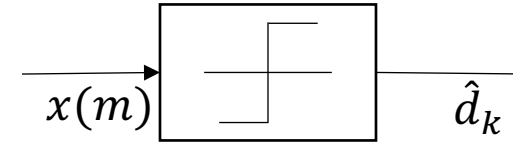
$$d(x(m), a^{(1)}) = |0.5 - 1| = 0.5$$

and will decide for the symbol that minimizes the distance in the signal space from the received sample  $x(m)$ , i.e.

$$\hat{a}_m = \arg \min_{a^{(i)} \in \mathcal{A}} d(x(m), a^{(i)}) = 1.$$

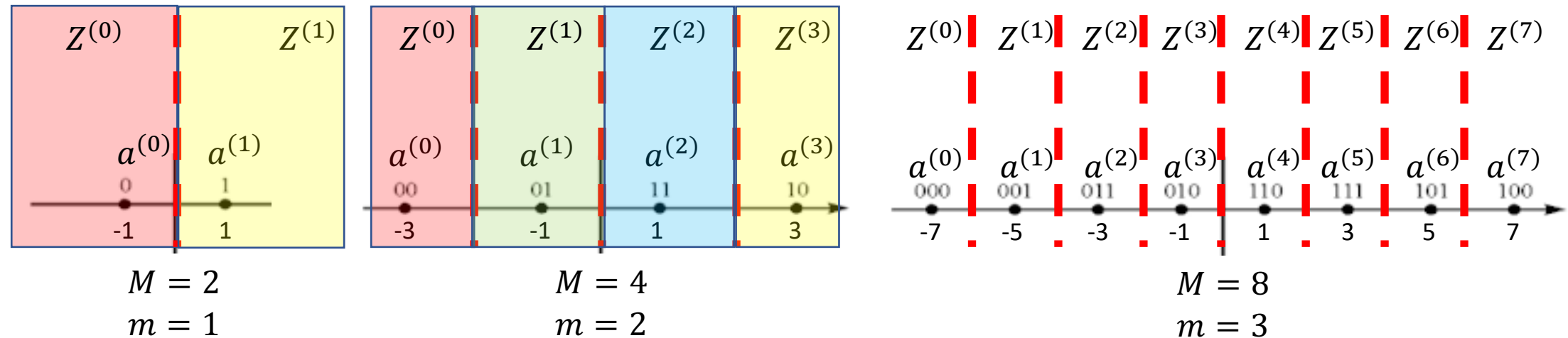


# Decision strategy



- Adopting the maximum likelihood criterion, we can partition the signal space in *zone of decisions*, where zone  $Z^{(i)}$  is the set of points that are closer to the symbol  $a^{(i)}$  than to any other symbol

$$Z^{(i)} = \{x | d(x, a^{(i)}) < d(x, a^{(j)}), j \neq i, j = 1, \dots, M\}$$



The decision threshold are in the midpoints of the segment connecting any two adjacent symbols. For example, for  $M = 4$  the thresholds are in  $-2, 0$  and  $2$ .

# PAM error probability

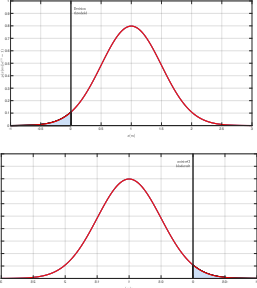
- Even if the maximum likelihood decision strategy is optimal, the receiver can still make some errors due to the presence of noise.
- The probability of error  $P(e|a^{(i)})$  is the probability that, having transmitted  $a^{(i)}$ , the decision variable  $x(m)$  due to the presence of noise does not fall in the decision region  $Z^{(i)}$ .
- The error probability is averaged over all the symbols of the constellation

$$P_e = \frac{1}{M} \sum_{i=0}^{M-1} P(e|a^{(i)}) = \lim_{N^{(s)} \rightarrow \infty} \frac{N_e^{(s)}}{N^{(s)}}$$

where  $N^{(s)}$  is the number of transmitted symbols and  $N_e^{(s)}$  is the number of symbol errors.

# PAM error probability: $Q$ -function

- The  $Q$ -function computes the integral of the *tail* of a Gaussian distribution.
- The probability that  $x \in \mathcal{N}(m_x, \sigma_x^2)$  is smaller than  $t_1 < m_x$  or larger than  $t_2 > m_x$  are the integral of Gaussian tails and they are computed as

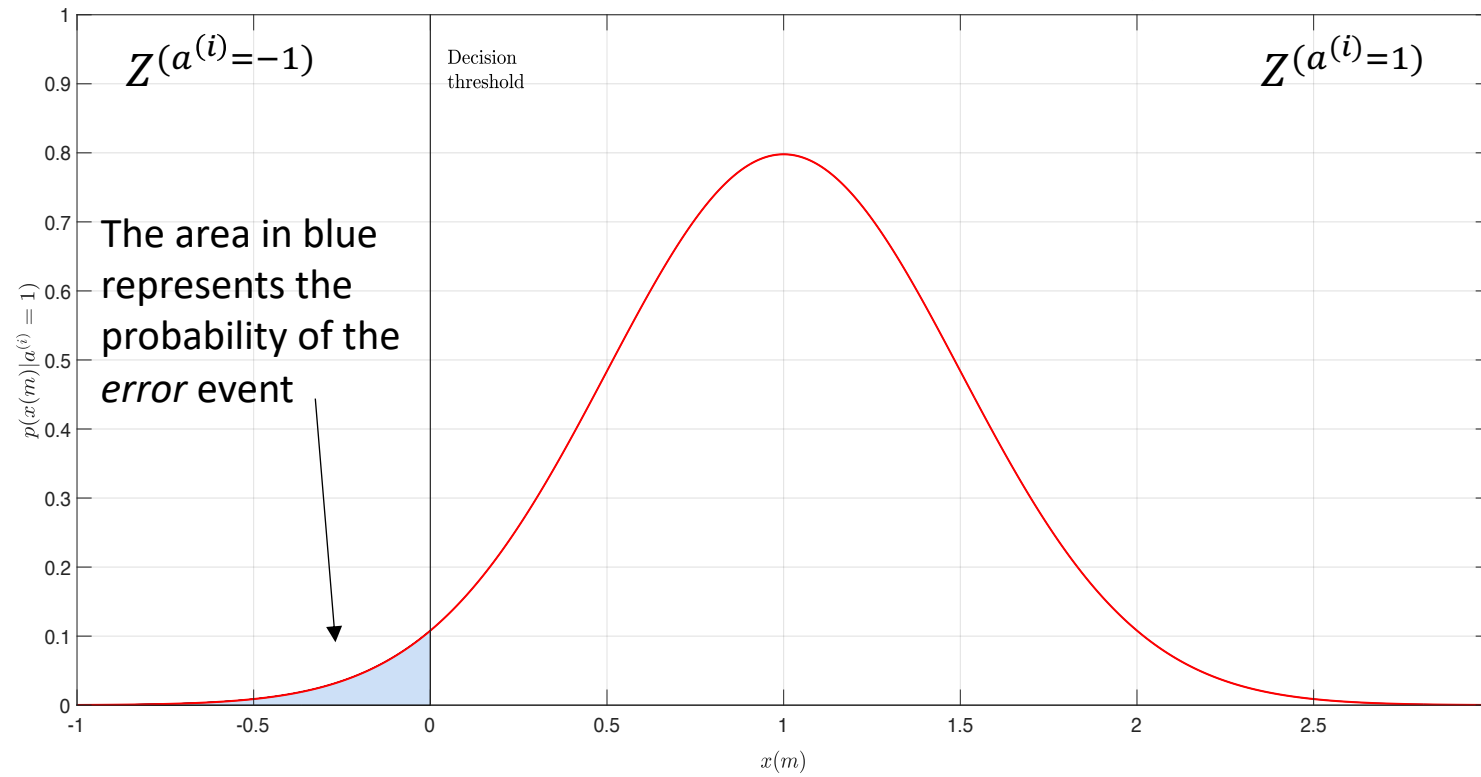

$$\left. \begin{aligned} \int_{-\infty}^{t_1} pdf(x) dx &= Q\left(\frac{m_x - t_1}{\sigma_x}\right) \\ \int_{t_2}^{+\infty} pdf(x) dx &= Q\left(\frac{t_2 - m_x}{\sigma_x}\right) \end{aligned} \right\} = Q\left(\frac{d(t_i, m_x)}{\sigma_x}\right), i = 1, 2$$

- In our case,  $m_x$  is the symbol  $a^{(i)}$  and  $t_1$  or  $t_2$  are the detection thresholds.
- The main properties of the  $Q$ -function are
$$Q(-\infty) = 1, Q(\infty) = 0, Q(0) = 0.5, Q(-x) = 1 - Q(x).$$

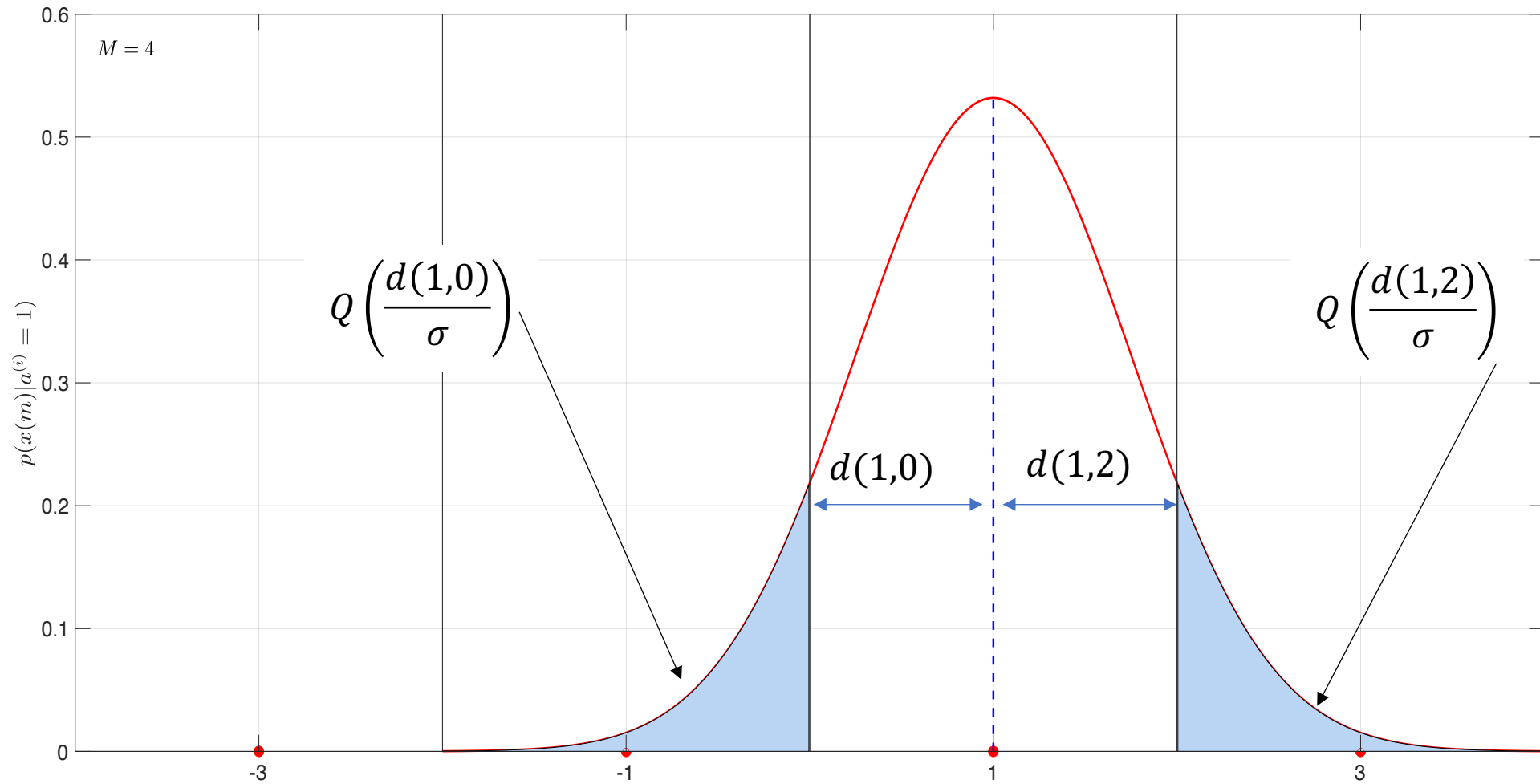
# PAM error probability ( $M = 2$ )

- To compute  $P(e|a^{(i)})$  we assume that  $x(m) = a^{(i)} + n(m)$  and the probability of error is

$$P(e|a^{(i)}) = \Pr\{x(m) \notin Z^{(i)} | a_m = a^{(i)}\} = Q\left(\frac{d(a^{(i)}, 0)}{\sigma}\right) = Q\left(\frac{1}{\sigma}\right)$$



# PAM error probability ( $M = 4$ )



$$P(e|a^{(i)} = 1) = \int_{-\infty}^0 p(x|a^{(i)} = 1)dx + \int_2^{+\infty} p(x|a^{(i)} = 1)dx = Q\left(\frac{d(1,0)}{\sigma}\right) + Q\left(\frac{d(1,2)}{\sigma}\right) = 2Q\left(\frac{1}{\sigma}\right)$$

# PAM error probability

- 2-PAM

$$P_e^{(2-PAM)} = \frac{1}{2} \left( Q \left( \frac{d(-1,0)}{\sigma} \right) + Q \left( \frac{d(1,0)}{\sigma} \right) \right) = Q \left( \frac{1}{\sigma} \right)$$

- 4-PAM

$$P_e^{(4-PAM)} = \frac{1}{4} \left( Q \left( \frac{d(-3,-2)}{\sigma} \right) + Q \left( \frac{d(-1,-2)}{\sigma} \right) + Q \left( \frac{d(-1,0)}{\sigma} \right) \right. \\ \left. Q \left( \frac{d(1,0)}{\sigma} \right) + Q \left( \frac{d(1,2)}{\sigma} \right) + Q \left( \frac{d(3,2)}{\sigma} \right) \right) = \frac{3}{2} Q \left( \frac{1}{\sigma} \right)$$

# PAM symbol error probability

- It is often useful to express the  $P_e$  in terms of  $E_s/N_0$ .

- 2-PAM:  $E_s = \frac{2^2-1}{6} = \frac{1}{2} \Rightarrow 2E_s = 1$  and  $\sigma^2 = N_0$ , and  $\frac{1}{\sigma} = \sqrt{\frac{1}{\sigma^2}} = \sqrt{\frac{2E_s}{N_0}}$ .

$$P_e^{(2-PAM)} = Q\left(\sqrt{\frac{2E_s}{N_0}}\right)$$

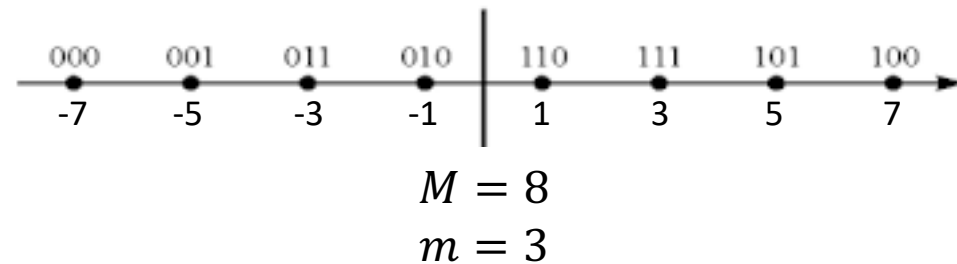
- 4-PAM:  $E_s = \frac{4^2-1}{6} = \frac{5}{2} \Rightarrow \frac{2}{5}E_s = 1$ , and  $\frac{1}{\sigma} = \sqrt{\frac{1}{\sigma^2}} = \sqrt{\frac{2E_s}{5N_0}}$ .

$$P_e^{(4-PAM)} = \frac{3}{2} Q\left(\sqrt{\frac{2E_s}{5N_0}}\right)$$

# Gray Mapping

- The way the  $m = \log_2 M$  bits are mapped on the  $M$  symbols of a PAM constellation is not unique.
- *Gray mapping* is a subset of all the possible mappings such that the string of  $m$  bits associated to adjacent symbols differ of only one bit.
- Example:  $M = 8, m = 3$

$a^{(0)} = -7 \rightarrow 000$   
 $a^{(1)} = -5 \rightarrow 00\mathbf{1}$   
 $a^{(2)} = -3 \rightarrow 0\mathbf{11}$   
 $a^{(3)} = -1 \rightarrow 0\mathbf{10}$   
 $a^{(4)} = 1 \rightarrow \mathbf{110}$   
 $a^{(5)} = 3 \rightarrow \mathbf{111}$   
 $a^{(6)} = 5 \rightarrow \mathbf{101}$   
 $a^{(7)} = 7 \rightarrow \mathbf{100}$





# PAM bit error probability

- To have a fair comparison, the modulation performance are expressed in terms of *bit error probability*  $P_e^{(b)}$  as function of  $E_b/N_0$ .
- The energy  $E_b$  per bit is computed as the energy per symbol divided by the number of bits per symbol

$$E_b = \frac{E_s}{\log_2 M}$$

- Although one symbol carries  $\log_2 M$  bits, it is reasonable to assume that in a well-designed system (*Gray mapping* and medium-high SNR) *a symbol error causes only one-bit errors*.
- If  $N^{(b)}$  and  $N_e^{(b)}$  are the number of transmitted bits and the number of bit errors, the bit error probability is computed as

$$P_e^{(b)} = \lim_{N^{(b)} \rightarrow \infty} \frac{N_e^{(b)}}{N^{(b)}} \approx \lim_{N^{(s)} \rightarrow \infty} \frac{N_e^{(s)}}{\log_2 M N^{(s)}} = \frac{1}{\log_2 M} \lim_{N^{(s)} \rightarrow \infty} \frac{N_e^{(s)}}{N^{(s)}} = \frac{1}{\log_2 M} P_e.$$

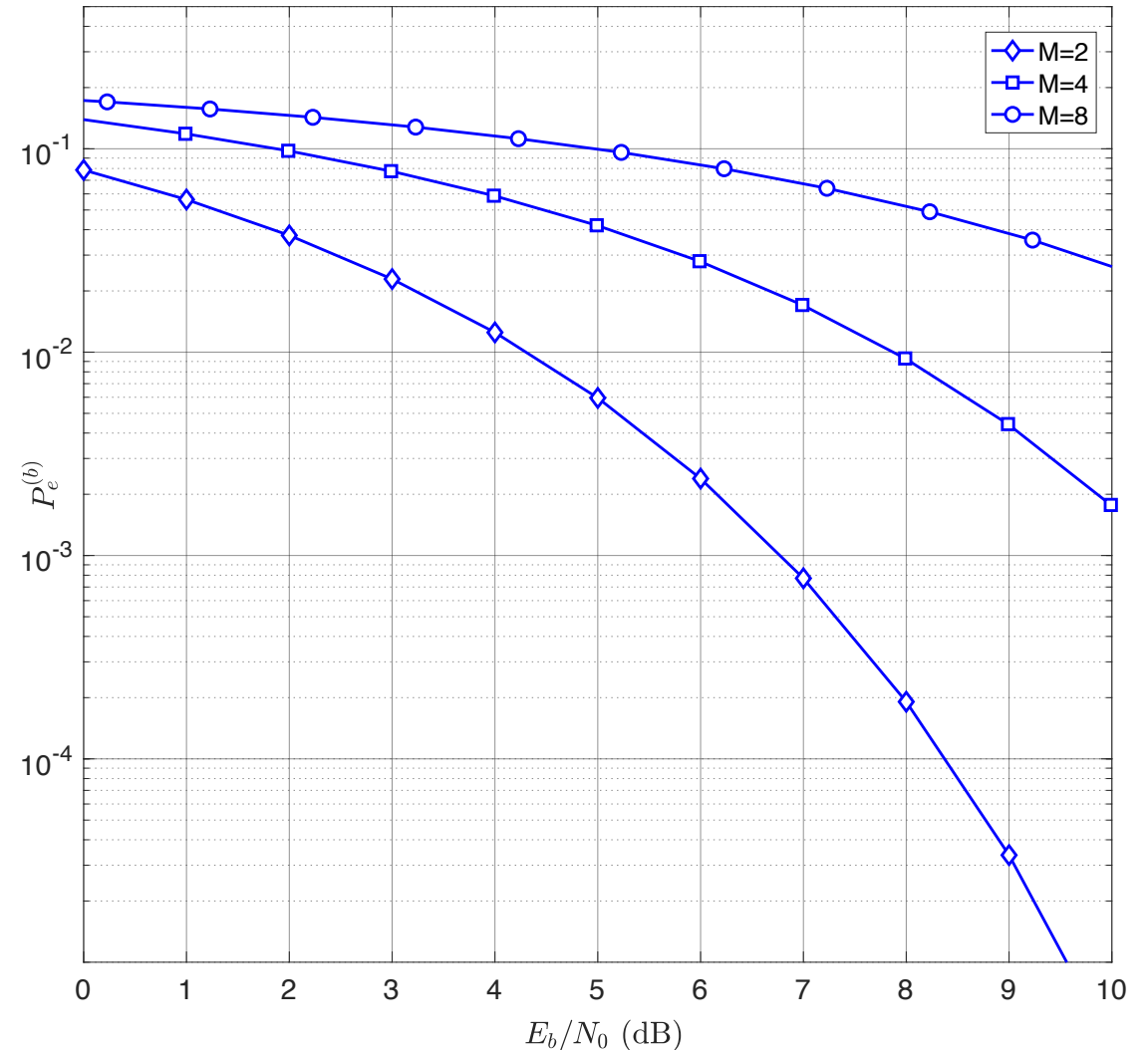
# PAM bit error probability

- 2-PAM:  $M = 2$ ,  $m = 1$  bit per symbol  $\Rightarrow P_e^{(b)} = P_e, E_b = E_s$

$$P_e^{(2-PAM),b} = Q\left(\sqrt{\frac{2E_b}{N_0}}\right)$$

- 4-PAM:  $M = 4$ ,  $m = 2$  bit per symbol  $\Rightarrow P_e^{(b)} = \frac{1}{2}P_e, E_b = \frac{1}{2}E_s$

$$P_e^{(4-PAM),b} = \frac{3}{4}Q\left(\sqrt{\frac{4E_b}{5N_0}}\right)$$



# Quadrature Amplitude Modulation (QAM)

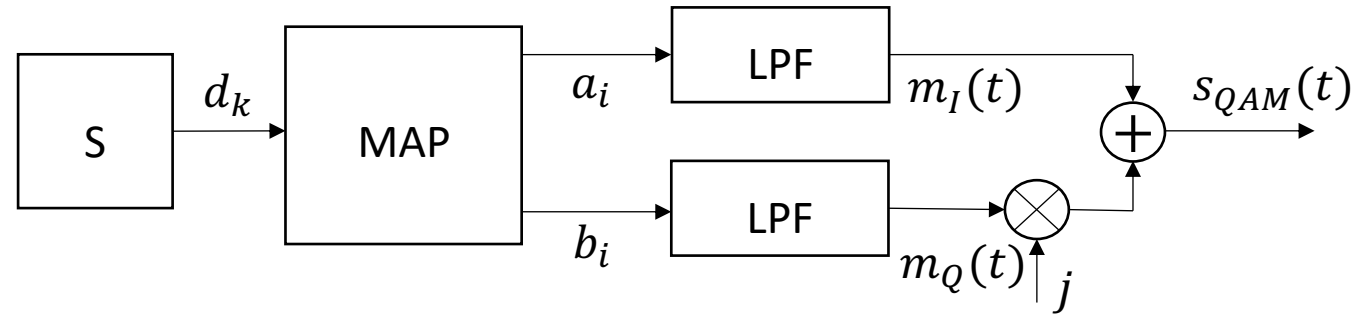
- In analog modulations, QAM is obtained by transmitting two orthogonal DSB signals  $m_I(t)$ ,  $m_Q(t)$  and the complex envelope is

$$\tilde{s}_{QAM}(t) = m_I(t) + jm_Q(t)$$

- Quadrature PAM (QAM) is obtained exactly in the same manner by transmitting two PAM signals in quadrature  $m_I(t) = \sum_i a_i g_T(t - iT)$  and  $m_Q(t) = \sum_i b_i g_T(t - iT)$ , with  $a_i, b_i$  PAM symbols.

$$s_{QAM}(t) = \sum_i a_i g_T(t - iT) + j \sum_i b_i g_T(t - iT)$$

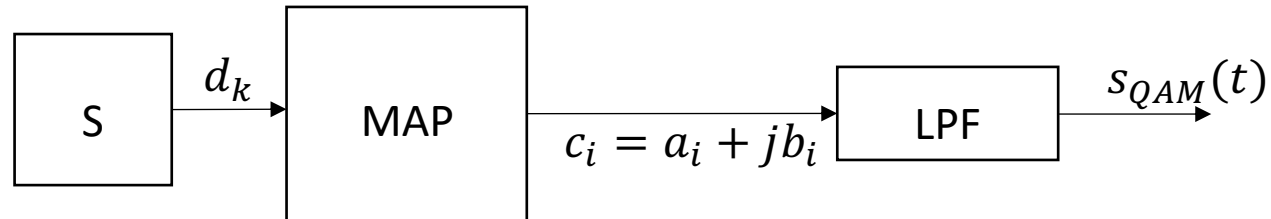
# QAM complex envelope



- The complex envelope of a QAM signal is

$$s_{QAM}(t) = \sum_i (a_i + j b_i) g_T(t - iT) = \sum_i c_i g_T(t - iT)$$

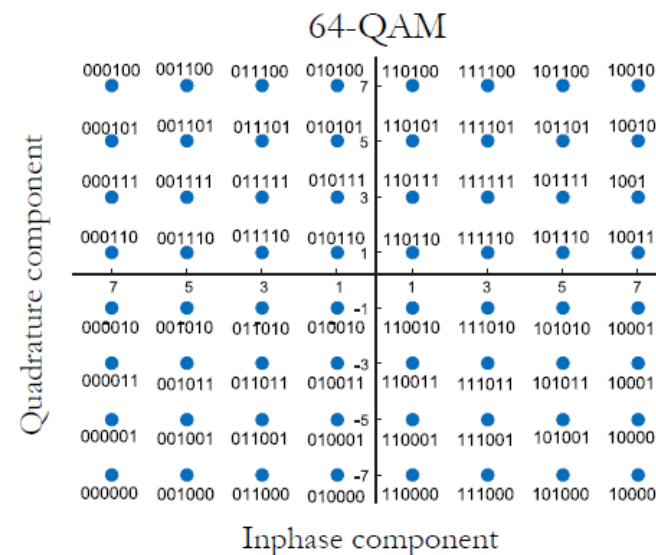
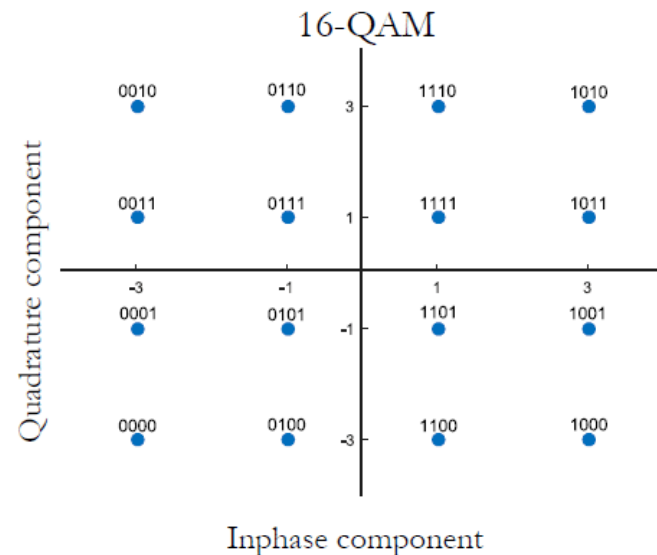
and the QAM *complex symbols* take the form  $c_i = a_i + j b_i$ .



- The only difference between PAM and QAM is the use of complex symbols!

# QAM symbols

- Since QAM is the combination of two orthogonal PAM with size  $M_{PAM}$ , the number of symbols is  $M_{QAM} = M_{PAM}^2$  and the number of transmitted bits is  $m_{QAM} = 2m_{PAM}$ , always even.
  - If the QAM is obtained by combining two 4-PAMs, then it is  $M_{QAM} = 16$  and  $m_{QAM} = 4$ , if  $M_{PAM} = 8$  then  $M_{QAM} = 64$  and  $m_{QAM} = 6$ .



# Energy of a QAM symbol

- In the computation of power and energy the only difference between PAM and QAM is in the mean square value of the symbols  $c_m$ .
- Keeping in mind that the in-phase and quadrature symbols are independent and zero-mean, it is

$$A = E\{c_m c_m^*\} = E\{a_m^2\} + E\{b_m^2\} = 2 \frac{M_{PAM}^2 - 1}{3} = 2 \frac{M_{QAM} - 1}{3}$$

- QAM constellation is much more compact and requires less energy per symbol.

$$A^{(4-PAM)} = \frac{16-1}{3} = 5; \quad A^{(4-QAM)} = 2 \frac{4-1}{3} = 2.$$

$$A^{(16-PAM)} = \frac{256-1}{3} = 85; \quad A^{(16-QAM)} = 2 \frac{16-1}{3} = 10.$$

- The energy per symbol is

$$E_s = \frac{A}{2} = \frac{M_{QAM} - 1}{3}$$

# QAM error probability

- The QAM *complex* decision variable is

$$\begin{aligned}x(m) &= c_m + n(m) = (a_m + jb_m) + (n_I(m) + jn_Q(m)) \\ &= a_m + n_I(m) + j(b_m + n_Q(m))\end{aligned}$$

- The in-phase and quadrature noise components  $n_I(m)$  and  $n_Q(m)$  are *independent*.
- Error events depend on noise. If the *noise is independent* also *the error events on the two components are independent*.

# Union bound

- The union of the sets A and B is computed as

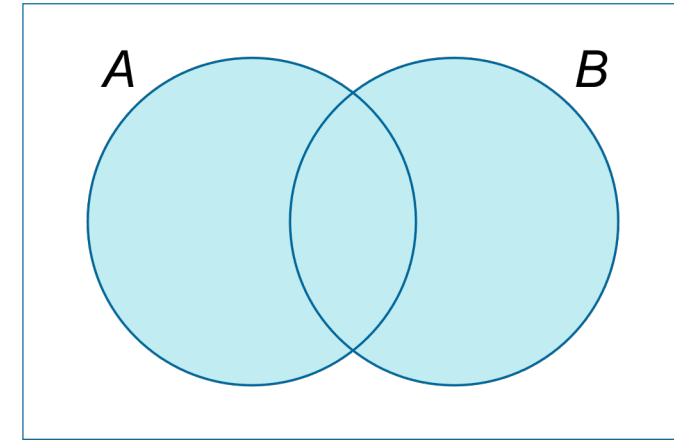
$$A \cup B = A + B - A \cap B$$

- In terms of probabilities, it is

$$\Pr\{A \cup B\} = \Pr\{A\} + \Pr\{B\} - \Pr\{A \cap B\}$$

- Since it is  $\Pr\{A \cap B\} \geq 0$ , a simple bound for  $\Pr\{A \cup B\}$  is the *union bound*

$$\Pr\{A \cup B\} \leq \Pr\{A\} + \Pr\{B\}$$





# QAM error probability – Union bound

- The event  $\mathcal{E}^{(i)} = \{\text{error} | c^{(i)}\}$  of making an error for a given transmitted symbol  $c^{(i)}$  can be computed as the union of the events

$$\mathcal{E}_I^{(i)} = \{\text{error on channel } I | c^{(i)}\} \text{ and } \mathcal{E}_Q^{(i)} = \{\text{error on channel } Q | c^{(i)}\}$$
$$\mathcal{E}^{(i)} = \mathcal{E}_I^{(i)} \cup \mathcal{E}_Q^{(i)}$$

- Applying the union bound, the error probability can be *upper bounded* by

$$P(e | c^{(i)}) = \Pr\{\mathcal{E}^{(i)}\} \leq \Pr\{\mathcal{E}_I^{(i)}\} + \Pr\{\mathcal{E}_Q^{(i)}\}$$

- Since the noise components are independent, the error probability can be computed independently on the two quadrature channels.
- Since M-QAM is obtained as the composition of two PAM in quadrature, each with  $\sqrt{M}$  symbols,

$$P_e^{(M\text{-QAM})} < 2P_e^{(\sqrt{M}\text{-PAM})}$$

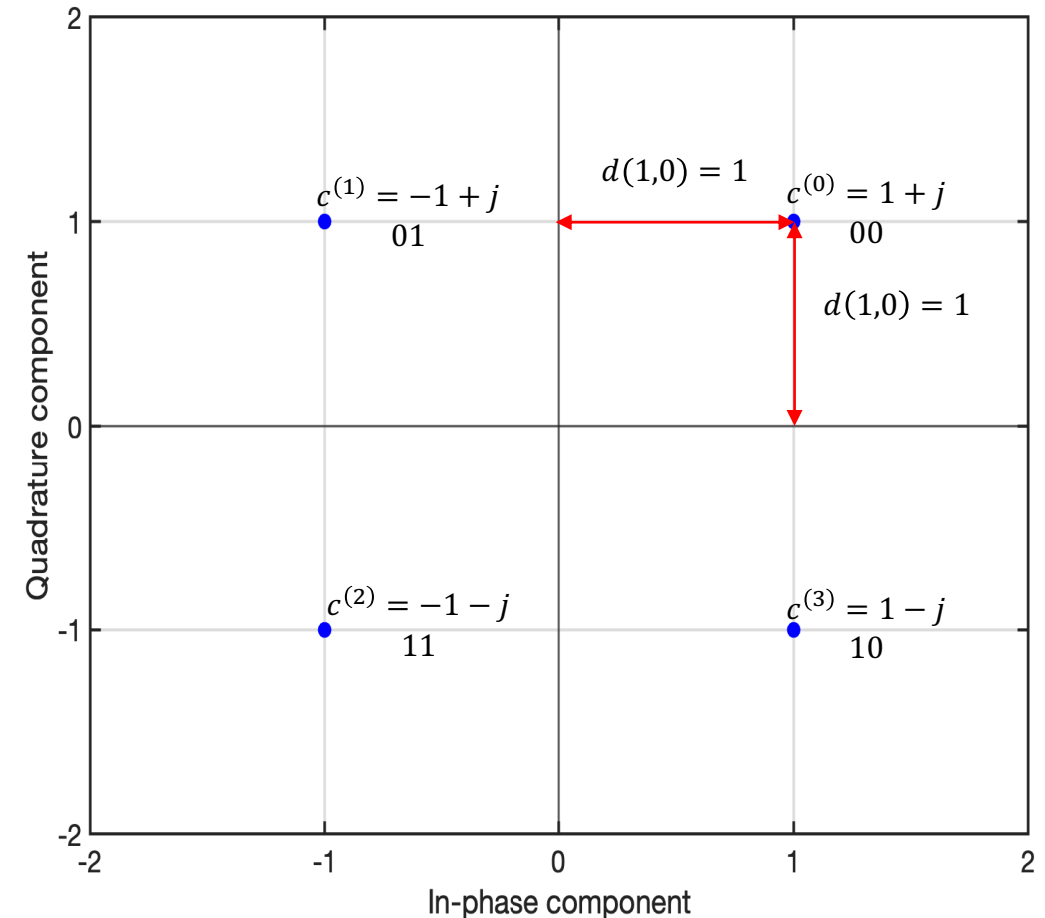
# 4-QAM error probability

- 4-QAM is obtained as the composition of two 2-PAM in quadrature.

- The symbol error probability is

$$P_e^{(4-QAM)} = \frac{1}{4} \sum_{i=0}^3 P(e|c^{(i)}) = P(e|c^{(0)})$$

$$\begin{aligned} P(e|c^{(0)}) &< Q\left(\frac{d(1,0)}{\sigma_{n_I}}\right) + Q\left(\frac{d(1,0)}{\sigma_{n_Q}}\right) \\ &= 2Q\left(\frac{1}{\sigma}\right) = 2P_e^{(2-PAM)} \end{aligned}$$



# QAM symbol error probability

- 4-QAM:  $M = 4, E_s = \frac{4-1}{3} = 1 \Rightarrow E_s = 1$ , and  $\frac{1}{\sigma} = \sqrt{\frac{1}{\sigma^2}} = \sqrt{\frac{E_s}{N_0}}$

$$P_e^{(4-QAM)} < 2Q\left(\sqrt{\frac{E_s}{N_0}}\right);$$

- 16-QAM:  $M = 16, E_s = \frac{16-1}{3} = 5 \Rightarrow \frac{1}{5}E_s = 1$ , and  $\frac{1}{\sigma} = \sqrt{\frac{1}{\sigma^2}} = \sqrt{\frac{E_s}{5N_0}}$

$$P_e^{(16-QAM)} < 2\frac{3}{2}Q\left(\frac{1}{\sigma}\right) = 3Q\left(\sqrt{\frac{E_s}{5N_0}}\right).$$

# M-QAM bit error probability

- The total number of M-QAM transmitted bits is the sum of the number of bits transmitted on the in-phase and quadrature channels, twice as many the bits transmitted by a PAM with  $\sqrt{M}$  symbols

$$m_{QAM} = \log_2 M = 2 \log_2 \sqrt{M}$$

- Accordingly,  $P_e^{(M-QAM,b)}$  can be computed as

$$\begin{aligned} P_e^{(M-QAM,b)} &= \lim_{N^{(b)} \rightarrow \infty} \frac{N_e^{(b)}}{N^{(b)}} \approx \lim_{N^{(s)} \rightarrow \infty} \frac{N_e^{(s)}}{\log_2 M N^{(s)}} = \frac{1}{\log_2 M} \lim_{N^{(s)} \rightarrow \infty} \frac{N_e^{(s)}}{N^{(s)}} = \frac{1}{\log_2 M} P_e^{(M-QAM)} \\ &= \frac{1}{2 \log_2 \sqrt{M}} 2 P_e^{(\sqrt{M}-PAM)} = P_e^{(\sqrt{M}-PAM,b)} \end{aligned}$$

- Although M-QAM carries twice as many bits, the bit error probability  $P_e^{(M-QAM,b)}$  is equal to the bit error probability of the corresponding  $\sqrt{M} - PAM$  modulation.

# QAM bit error probability

- 4-QAM:

$$\frac{1}{\sigma^2} = \frac{E_s}{N_0} = \frac{2E_b}{N_0}$$

$$P_e^{(4-QAM,b)} = P_e^{(2-PAM,b)} = Q\left(\frac{1}{\sigma}\right) = Q\left(\sqrt{\frac{2E_b}{N_0}}\right)$$

- 16-QAM

$$\frac{1}{\sigma^2} = \frac{E_s}{5N_0} = \frac{4E_b}{5N_0}$$

$$P_e^{(16-QAM,b)} = P_e^{(4-PAM,b)} = \frac{3}{4} Q\left(\frac{1}{\sigma}\right) = \frac{3}{4} Q\left(\sqrt{\frac{4E_b}{5N_0}}\right)$$

