

Telecommunicatie A (EE2T11)

Lecture 7 overview:

Transmission of digital signals in baseband

- * **Line codes**
- * **Power spectra**
- * **Transmission quality: eye-pattern, regenerative repeater**
- * **Spectral efficiency**

Inter-symbol interference (ISI)

- * **Pulse shaping (Nyquist criterion)**
- * **Sinc-pulses vs. Raised Cosine pulses**

EE2T11 Telecommunicatie A

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Colleges en Werkcolleges Telecommunicatie A

Colleges:

Dinsdag 29-3 7^e en 8^e uur, EWI-Pi

Werkcolleges:

Maandag 4-4 5^e en 6^e uur, EWI-Pi

Review digital signaling (1)

A waveform of N symbols (N dimensions) of a digital signal can be written as a sequence of N orthogonal terms:

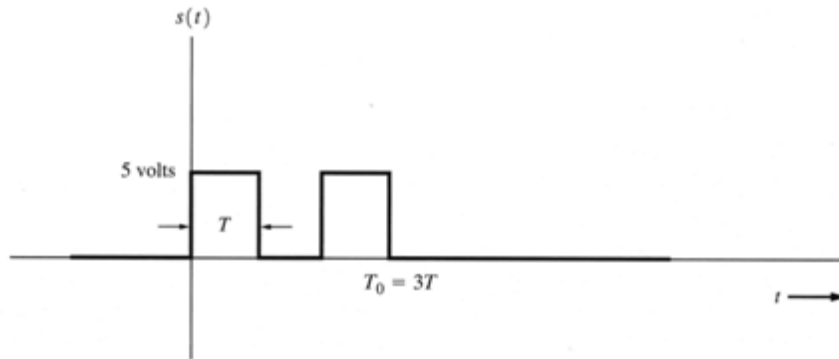
$$w(t) = \sum_{k=1}^N w_k \varphi_k(t) \quad 0 \leq t \leq T_0$$

w_k represents the k th symbol value and contains the digital information.

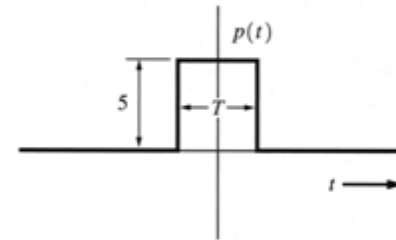
$\varphi_k(t)$ are orthogonal functions (in general analog waveforms), like a time slot, frequency, phase or a code, with:

$$\int \varphi_k(t) \varphi_l^*(t) dt = \begin{cases} 1 & \text{for } k = l \\ 0 & \text{for } k \neq l \end{cases} \quad k = 1, 2, \dots, N$$

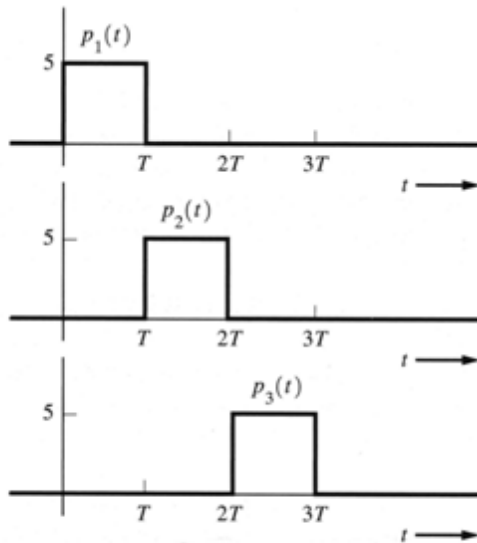
Review digital signaling (2)



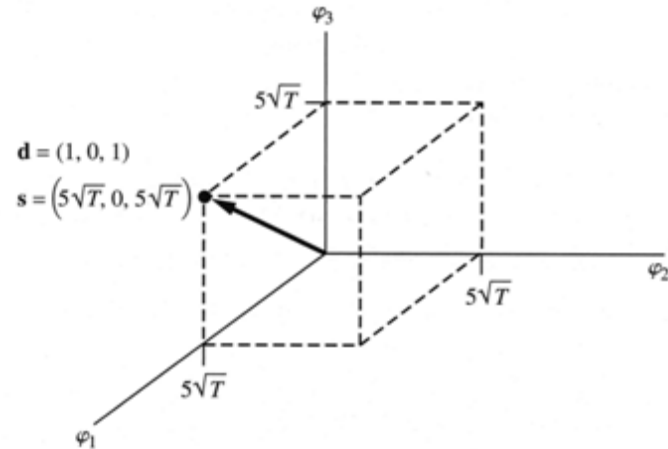
(a) A Three-Bit Signal Waveform



(b) Bit Shape Pulse



(c) Orthogonal Function Set



(d) Vector Representation of the 3-Bit Signal

Figure 3-11 Representation for a 3-bit binary digital signal.

Review digital signaling (3)

The N functions $\varphi_k(t)$ ($k = 1, \dots, N$) span an N -dimensional space. A message of N symbols can be represented by a vector in this N -dimensional space:

$$\underline{w} = \sum_{k=1}^N w_k \varphi_k$$

How do we detect the data after transmission over a channel?
By correlation with the complex conjugate of the orthogonal functions $\varphi_k(t)$:

$$w_k = \int_0^{T_0} \underline{w} \cdot \varphi_k^*(t) dt \quad \text{for } k = 1, 2, \dots, N$$

This is optimal detection for a channel with Additive White Gaussian Noise (AWGN): "matched-filter" detector.

Baseband transmission

Transmission of digital baseband signals: usually **line transmission**.

$$w(t) = \sum_{k=-\infty}^{\infty} w_k \varphi(t - kT_s) \quad \text{with } T_s \text{ the symbol time.}$$

How do we transmit a "1" or a "0" over a line?

Usually in a serial manner, but different formats exist:

line codes

There are two main categories of *line codes*:

- a) Non-Return-to-Zero (NRZ)
- b) Return-to-Zero (RZ)

Line codes

Different line codes:

- punch tape code
- Unipolar NRZ (On-Off-Keying, OOK)
- Polar NRZ
- Unipolar RZ
- Bipolar RZ
- Manchester RZ
(or pseudo ternary
or AMI (Alternate Mark Inversion))
- Manchester coding
(or split phase coding)

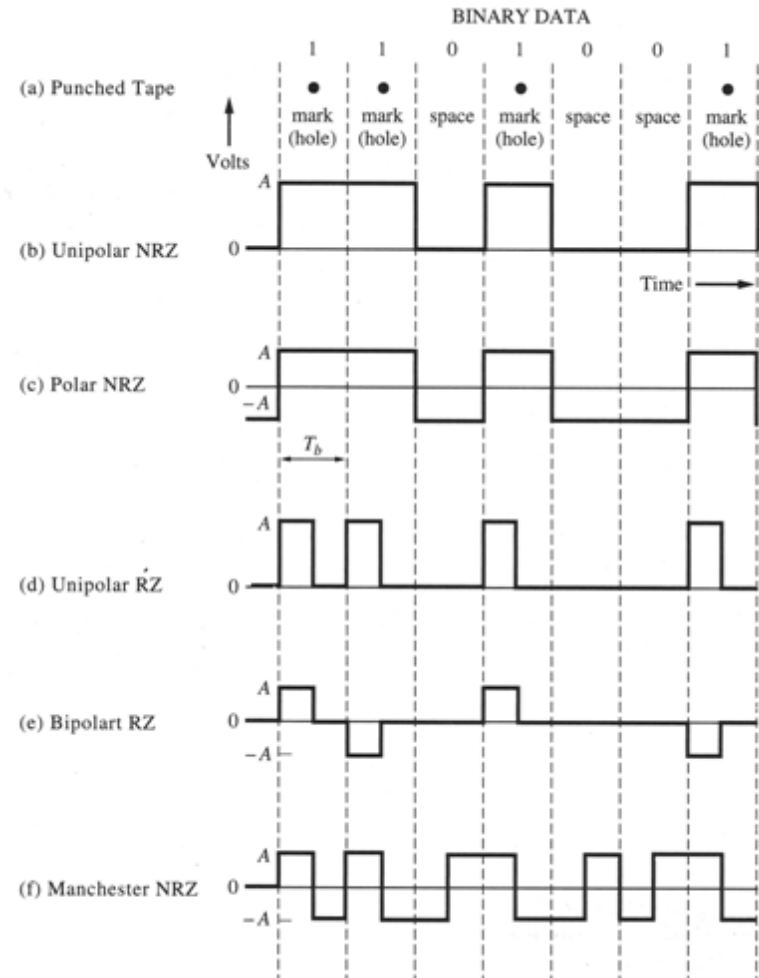


Figure 3-15 Binary signaling formats.

Features of line codes

Desirable features of *line codes*:

1. Spectrum adapted to the *available* baseband channel
2. Self-synchronizing (timing information available also for long sequences of identical symbols)
3. Low bit error probability
4. Small bandwidth
5. Build-in error detection -correction
6. Transparent and unambiguous

Power spectra line codes (1)

For a general line code:

$$s(t) = \sum_{n=-\infty}^{\infty} a_n f(t - nT_s) = f(t) * \sum_{n=-\infty}^{\infty} a_n \delta(t - nT_s)$$

In general, the *Power Spectral Density* is defined as:

$$P_s(f) = \lim_{T \rightarrow \infty} \left(\frac{|S_T(f)|^2}{T} \right) = \mathfrak{F}\{R_s(\tau)\}$$

with: $R_s(\tau) = \langle s(t)s(t+\tau) \rangle = \lim_{T \rightarrow \infty} \frac{1}{T} \int_{-T/2}^{T/2} s(t)s(t+\tau) dt$

Power Spectral Density (PSD) of a digital signal waveform (6.70):

$$P_s(f) = \frac{|F(f)|^2}{T_s} \left[\sum_{k=-\infty}^{+\infty} R(k) e^{-j\omega k T_s} \right]$$

Power spectra line codes (2)

For the *PSD* of a general line code, two factors are important:

* Pulse shape: $F(f)$ is the amplitude spectrum of waveform $f(t)$.

* Data:
$$R(k) = \overline{a_n a_{n+k}} = \sum_{i=1}^I (a_n a_{n+k})_i P_i$$

where P_i is the probability of the i^{th} outcome of the product $a_n a_{n+k}$, which has I possibilities.

We use here *stochastic techniques* because we observe a sequence of random bits or symbols with equal probability.

The full derivation of the *PSD* of a line code signal is given in Chapter 6.2.

Power spectrum Polar NRZ

The signal $s(t) = \sum_{n=-\infty}^{\infty} a_n f(t - nT_b)$ with $a_n \in \{-A, +A\}$ uses

square pulses $f(t) = \text{rect}\left(\frac{t}{T_b}\right) \Rightarrow F(f) = T_b \text{sinc}(fT_b) = T_b \frac{\sin(\pi fT_b)}{\pi fT_b}$

Since: $R(k=0) = \overline{a_n^2} = A^2$: $I=1, a_n^2 = A^2, P_i=1$

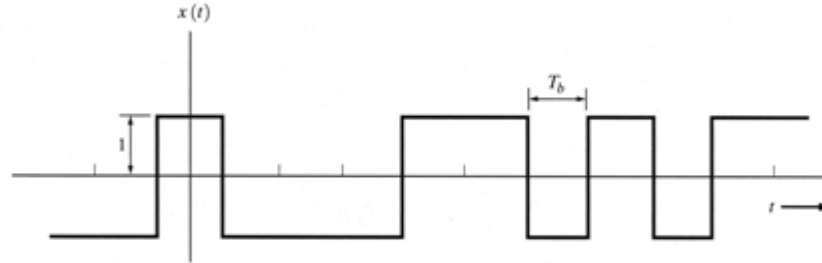
$R(k \neq 0) = \overline{a_n a_{n+k}} = 0$: $I=2, a_n a_{n+k} \in \{-A^2, A^2\}, P_i=1/2$

Now it follows:

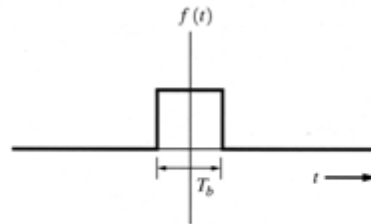
$$P_{P-NRZ}(f) = \frac{|F(f)|^2}{T_b} \left[\sum_{k=-\infty}^{+\infty} R(k) e^{-j\omega k T_b} \right] = \frac{A^2}{T_b} |F(f)|^2 = A^2 T_b \text{sinc}^2(fT_b)$$

with bit rate $R_b = \frac{1}{T_b}$.

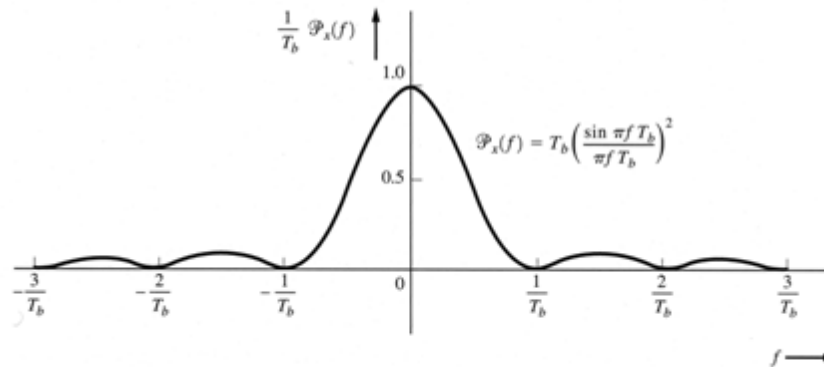
Power spectrum Polar NRZ (2)



(a) Polar Signal



(b) Signaling Pulse Shape



(c) Power Spectral Density of a Polar Signal

Figure 6-5 Random polar signal and its PSD.

Power spectrum Uni-polar NRZ (1)

In this case $a_n \in \{0, A\}$ and now we find for the autocorrelation of the data:

$$R(k) = \overline{a_n a_{n+k}} = \sum_{i=1}^I (a_n a_{n+k})_i P_i$$

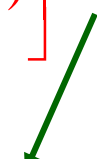
$$k = 0 \rightarrow a_n^2 = \begin{cases} A^2 & \text{with prob. } 1/2 \\ 0 & \text{with prob. } 1/2 \end{cases} \Rightarrow R(0) = \frac{A^2}{2}$$

$$k \neq 0 \rightarrow a_n a_{n+k} = \begin{cases} A \cdot A = A^2 & \text{with prob. } 1/4 \\ 0 \cdot A = 0 & \text{with prob. } 1/4 \\ A \cdot 0 = 0 & \text{with prob. } 1/4 \\ 0 \cdot 0 = 0 & \text{with prob. } 1/4 \end{cases} \Rightarrow R(k \neq 0) = \frac{A^2}{4}$$

Power spectrum Unipolar NRZ (2)

Again we use square pulses. Using (3-36a), we find for the PSD:

$$\begin{aligned} P_{U-NRZ}(f) &= \frac{|F(f)|^2}{T_b} \left[\sum_{k=-\infty}^{+\infty} R(k) e^{-j\omega k T_b} \right] \\ &= T_b \frac{\sin^2(\pi f T_b)}{(\pi f T_b)^2} \left[\frac{A^2}{4} + \sum_{k=-\infty}^{\infty} \frac{A^2}{4} e^{-2\pi j k f T_b} \right] = \frac{A^2 T_b}{4} \text{sinc}^2(f T_b) \left[1 + \sum_{k=-\infty}^{\infty} e^{-2\pi j k f T_b} \right] \\ &= \frac{A^2 T_b}{4} \text{sinc}^2(f T_b) \left[1 + \frac{1}{T_b} \sum_{n=-\infty}^{\infty} \delta\left(f - \frac{n}{T_b}\right) \right] = \frac{A^2 T_b}{4} \text{sinc}^2(f T_b) \left[1 + \frac{1}{T_b} \delta(f) \right] \end{aligned}$$


$$\frac{\sin \pi f T_b}{\pi f T_b} = \frac{\sin n\pi}{n\pi} = 0 \text{ at } f = \frac{n}{T_b} \text{ for } n \neq 0$$

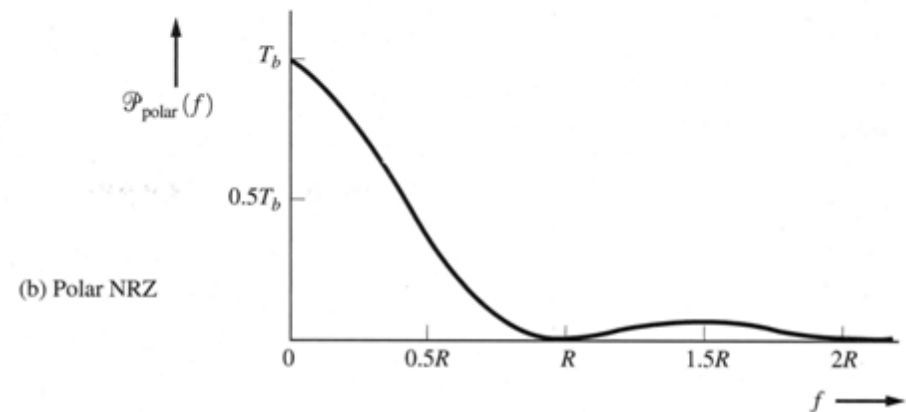
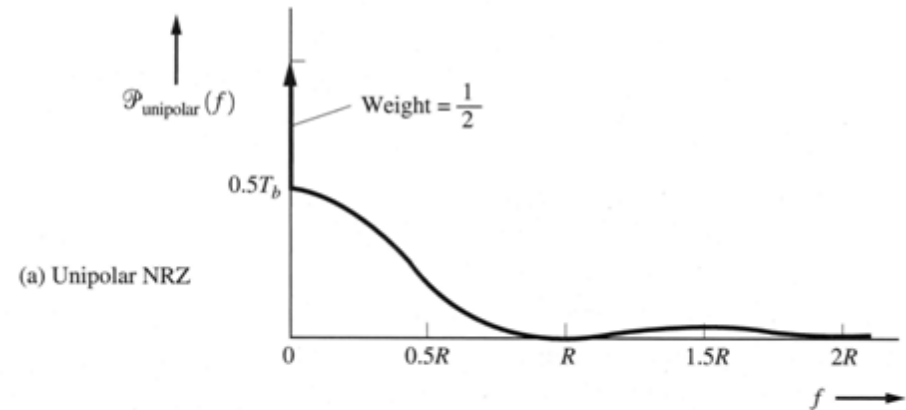
Compare Polar NRZ and Unipolar NRZ

$$P_{\text{Unipolar NRZ}}(f) = \frac{A^2 T_b}{4} \text{sinc}^2(fT_b) \left[1 + \frac{1}{T_b} \delta(f) \right]$$

For a normalized
power $P = 1 \rightarrow A = \sqrt{2}$
Due to a DC-level of $A/2$,
power is wasted.

$$P_{\text{Polar NRZ}}(f) = A^2 T_b \text{sinc}^2(fT_b)$$

For a normalized
power $P = 1 \rightarrow A = 1$




Power spectrum Unipolar RZ (1)

Square pulses: $f(t) = \Pi\left(\frac{2t}{T_b}\right) \Rightarrow F(f) = \frac{T_b}{2} \text{sinc}\left(\frac{fT_b}{2}\right)$

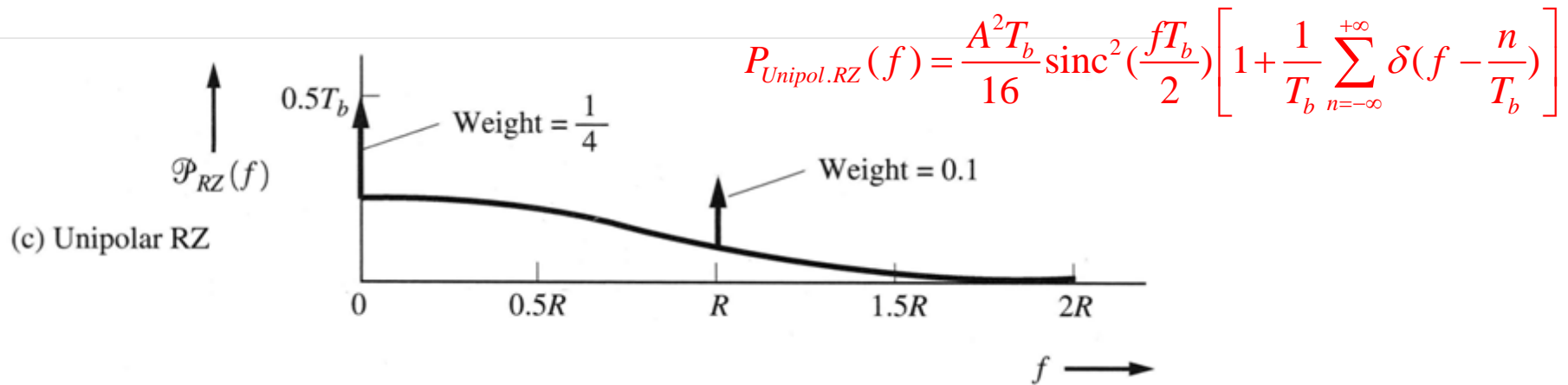
Using (3-36a), we find for the PSD:

$$\begin{aligned} P_{Unipol.RZ}(f) &= \frac{A^2 T_b}{16} \text{sinc}^2\left(\frac{fT_b}{2}\right) \left[1 + \sum_{k=-\infty}^{+\infty} e^{j\omega k T_b} \right] \\ &= \frac{A^2 T_b}{16} \text{sinc}^2\left(\frac{fT_b}{2}\right) \left[1 + \frac{1}{T_b} \sum_{n=-\infty}^{+\infty} \delta\left(f - \frac{n}{T_b}\right) \right] \end{aligned}$$

For power $P = 1 \rightarrow A = 2$

$$\frac{\sin \frac{\pi}{2} f T_b}{\frac{\pi}{2} f T_b} = \frac{\sin n \frac{\pi}{2}}{n \frac{\pi}{2}} = 0 \text{ at } f = \frac{n}{T_b} \text{ for } n = 2m \neq 0$$


Power spectrum Unipolar RZ (2)



- an infinite train of δ -functions at odd multiples of R !
- spectrum twice as wide as for NRZ
- synchronization simple, because of the δ -function at the clock frequency $f = 1/T_b = R_b$

Bipolar RZ and Manchester NRZ

Power spectra:

$$P_{Bip.RZ}(f) = \frac{A^2 T_b}{4} \text{sinc}^2\left(\frac{fT_b}{2}\right) \sin^2 \pi f T_b$$

For power $P = 1 \rightarrow A = 2$

$$P_{Manchester}(f) = A^2 T_b \text{sinc}^2\left(\frac{fT_b}{2}\right) \sin^2 \frac{\pi f T_b}{2}$$

For power $P = 1 \rightarrow A = 1$

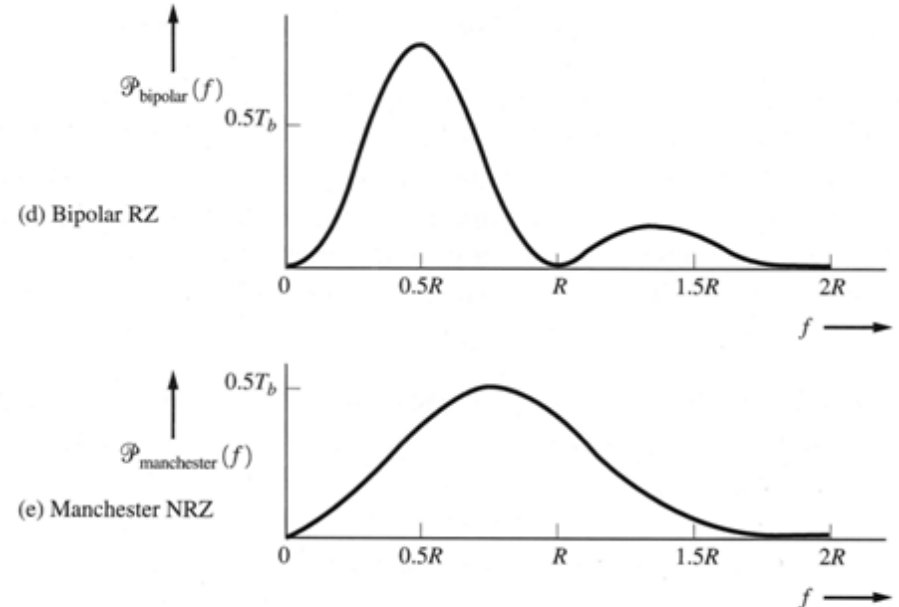


Figure 3-16 PSD for line codes (positive frequencies shown).

Synchronization Bipolar-RZ?

Line codes

Different line codes:

- punch tape code
- Unipolar NRZ (On-Off-Keying, OOK)
- Polar NRZ
- Unipolar RZ
- Bipolar RZ
- Manchester coding
(or split phase coding)

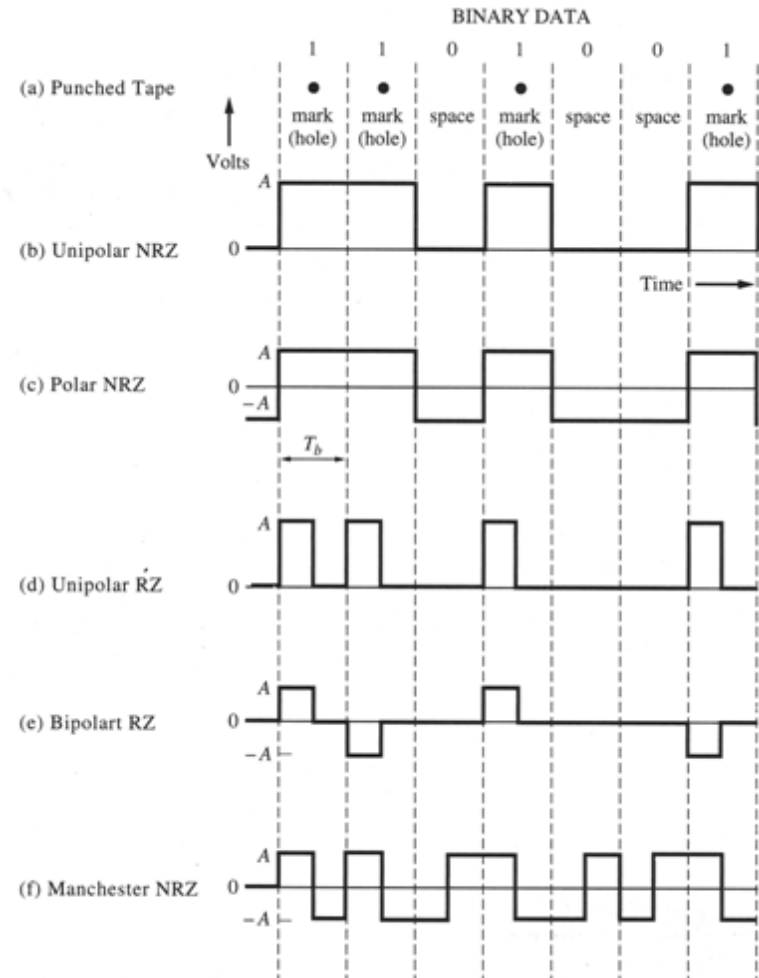


Figure 3-15 Binary signaling formats.

Bipolar RZ and Manchester NRZ

Features:

- no DC-component
- no/little energy around $f = 0$
- synchronization Bip. RZ based on rectified signal
- Bi-polar RZ has sync. problems for long sequences of zeros (HDBn)
- $BW_{\text{Manchester}} > BW_{\text{Bip. RZ}}$

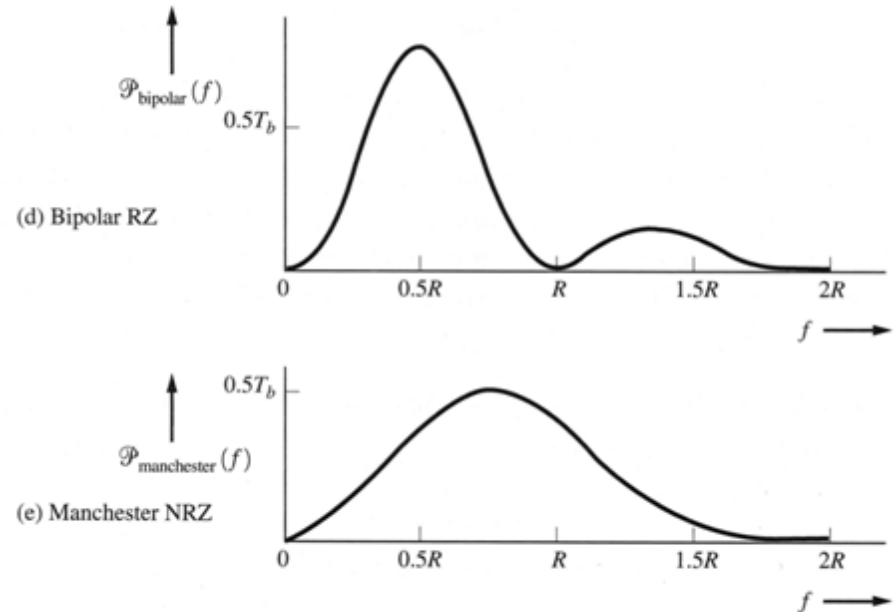
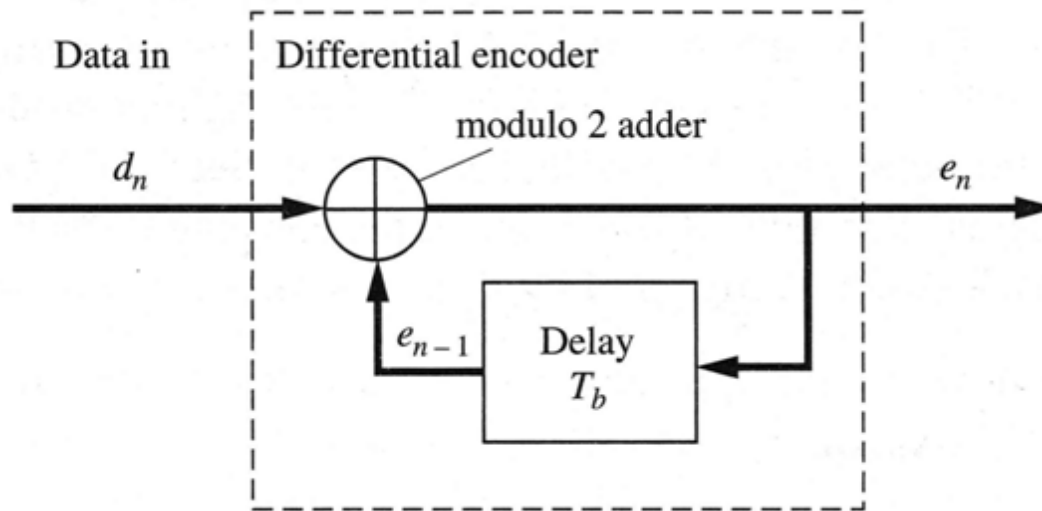


Figure 3-16 PSD for line codes (positive frequencies shown).

Differential coding

Polarity errors result in: "1" \leftrightarrow "0". This can be prevented by

differential coding: $e_n = d_n \oplus e_{n-1}$ (modulo 2 addition, XOR).



| d_n | e_{n-1} | e_n |
|-------|-----------|-------|
| 0 | 0 | 0 |
| 0 | 1 | 1 |
| 1 | 0 | 1 |
| 1 | 1 | 0 |

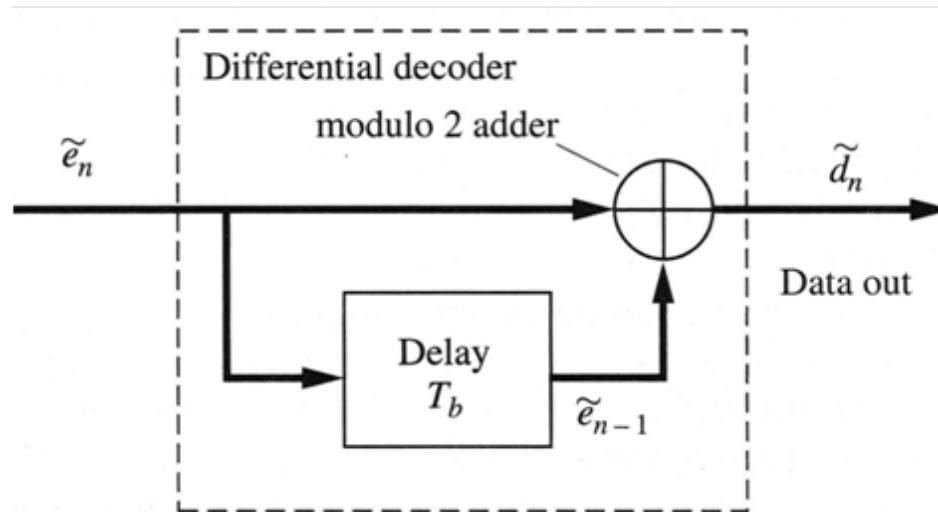
A "1" in the data signal changes the sign of the line signal.

Differential decoding

Decoding:

$$\hat{d}_n = e_n \oplus e_{n-1} \Rightarrow$$

A change of sign → "1";
No change of sign → "0".



The information is *not coded in the symbol values,*
but in the transitions.

Differential coding - decoding

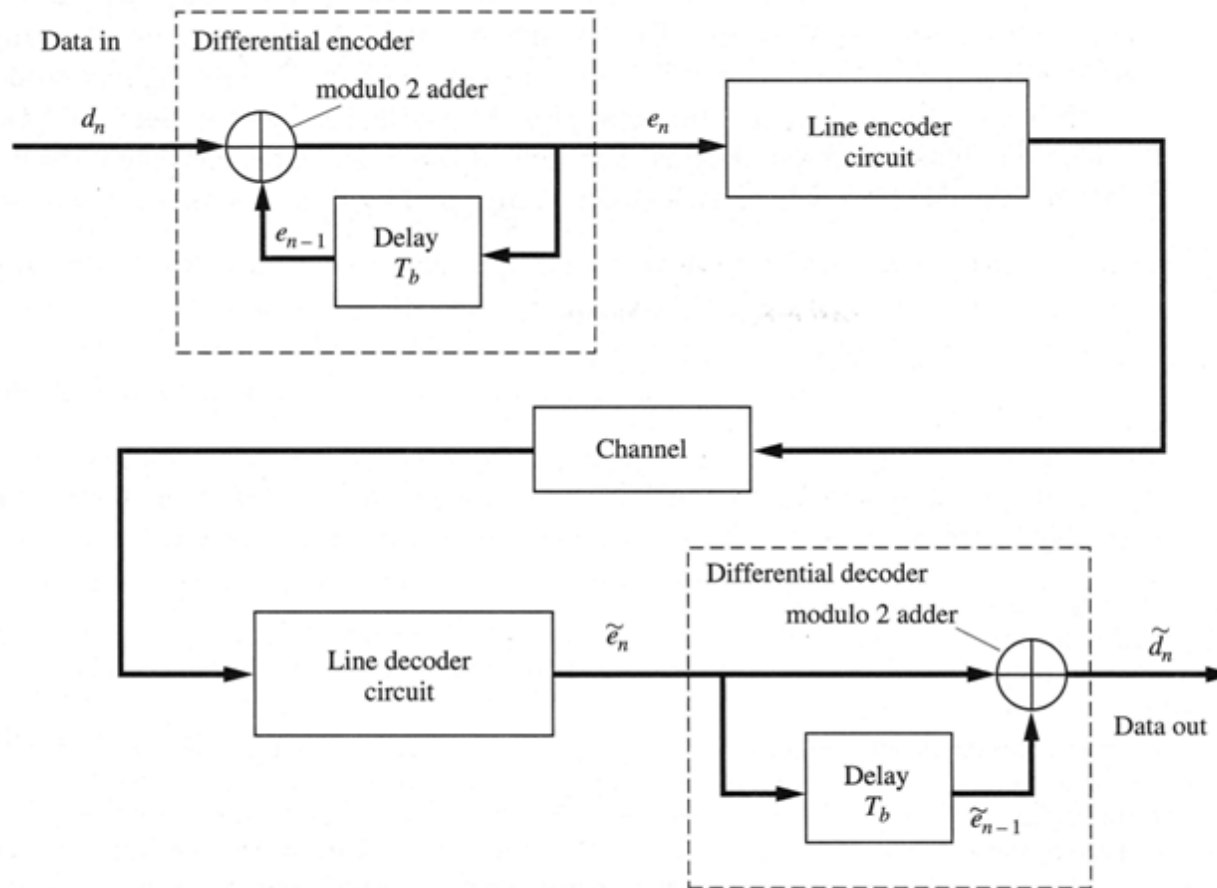


Figure 3-17 Differential coding system.

Example: differential coding (1)

Data sequence :

$d =$ 1 0 0 1 1 0 1 1 1 0 0 1

Differential encoded sequence:

or Reference symbol

$e =$ 0 1 1 1 0 1 1 0 1 0 0 0 1

$e =$ 1 0 0 0 1 0 0 1 0 1 1 1 0

In both cases, with and without inversion the original sequence is obtained after decoding:

$\hat{d} =$ 1 0 0 1 1 0 1 1 1 0 0 1

Example: differential coding (2)

Data sequence :

$$d = \quad 1 \ 0 \ 0 \ 1 \ 1 \ 0 \ 1 \ 1 \ 1 \ 0 \ 0 \ 1$$

What happens when **ERRORS** occur in the reception of the encoded sequence:

$$\begin{array}{l} e = 0 \ 1 \ 1 \ 1 \ 0 \ 1 \ 1 \ 0 \ 1 \ 0 \ 0 \ 0 \ 1 \\ \text{Reference symbol} \swarrow \searrow \\ \hat{e} = 0 \ 1 \ 1 \ 0 \ 0 \ 1 \ 1 \ 0 \ 0 \ 1 \ 0 \ 0 \ 1 \end{array}$$

Sequence is obtained after decoding:

$$\hat{d} = \quad 1 \ 0 \ 1 \ 0 \ 1 \ 0 \ 1 \ 0 \ 1 \ 1 \ 0 \ 1$$

Every **single or burst** of errors results in **two decoding errors**.

Transmission quality: Eye pattern

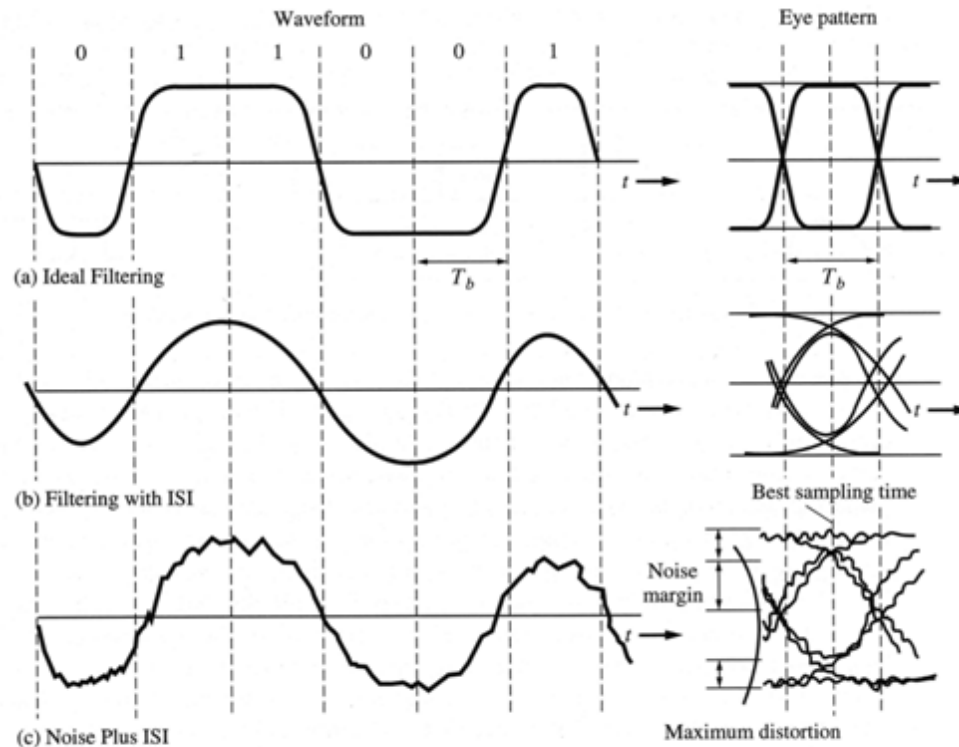


Figure 3-18 Distorted polar NRZ waveform and corresponding eye pattern.

Signal distortion and noise result in closure of the eye: in width and height.

The effect of filtering (rounding and dispersion of pulses) and noise (distortion) can be observed in the eye-pattern. For a good detection quality, the eye-pattern should be open as far as possible!

PCM System

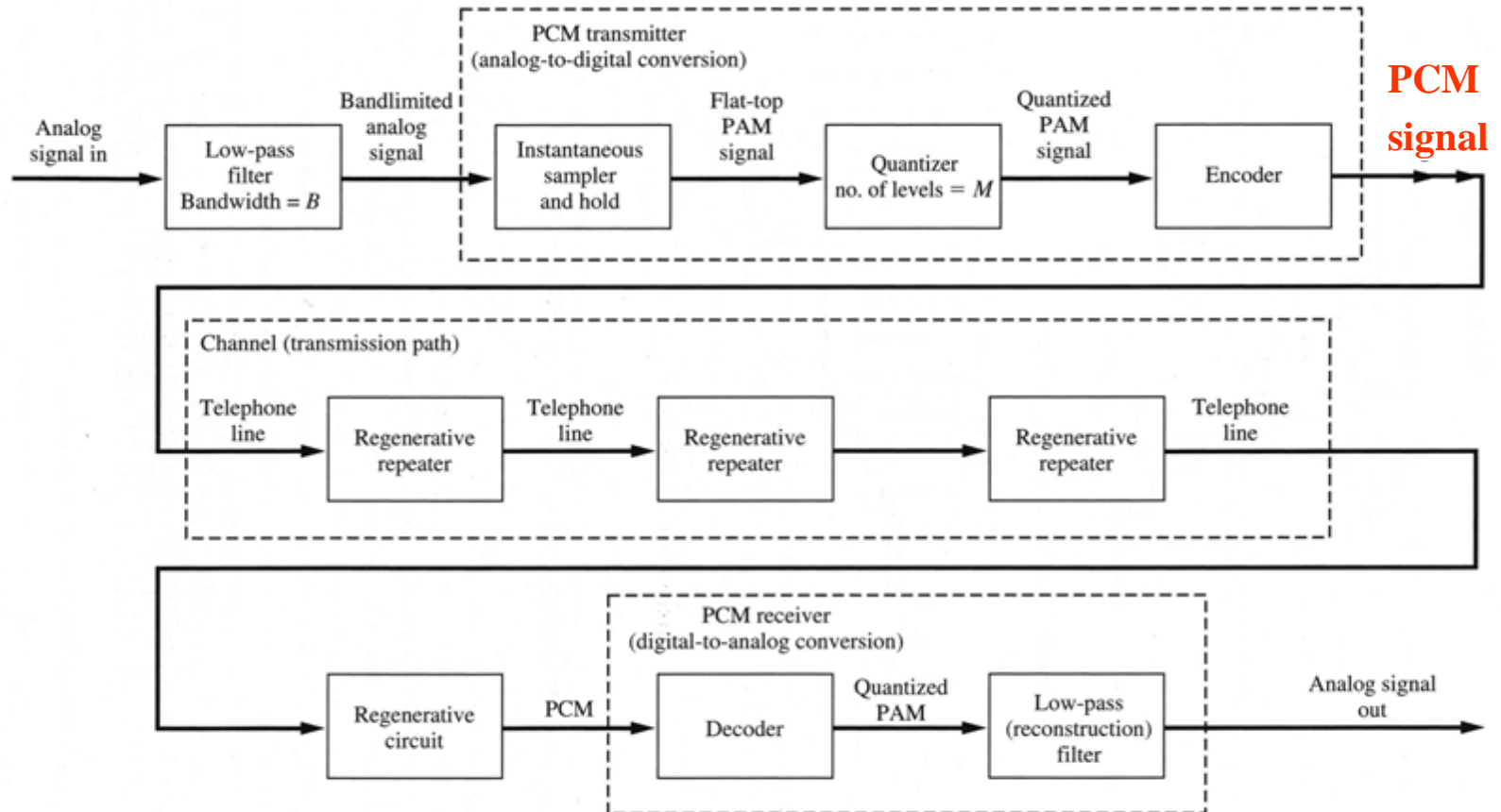


Figure 3-7 PCM transmission system.

Regenerative repeater

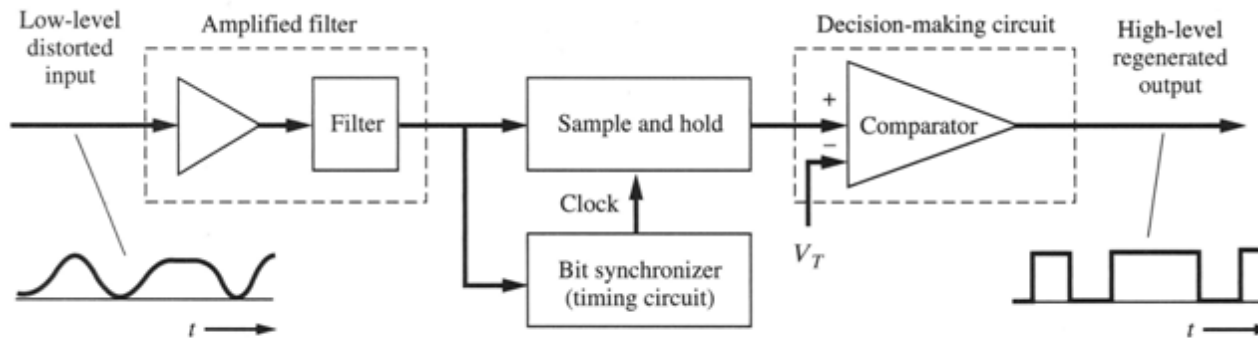


Figure 3-19 Regenerative repeater for unipolar NRZ signaling.

In a regenerative repeater, degenerated pulses due to filtering effects and noise are restored after detection at the cost of occasional errors.

With m repeaters the probability of i errors is: $P_i = \binom{m}{i} P_e^i (1 - P_e)^{m-i}$

However, only an odd number of errors really result in a detection error in the end (for binary signals), so:

$$P_{me} = \sum_{i=1,3,5,\dots} P_i = \sum_{i=1,3,5,\dots} \binom{m}{i} P_e^i (1 - P_e)^{m-i} \approx m P_e (1 - P_e)^{m-1}$$

$$\text{thus for } P_e \ll 1 \Rightarrow P_{me} \approx m P_e$$

Multilevel signaling (1)

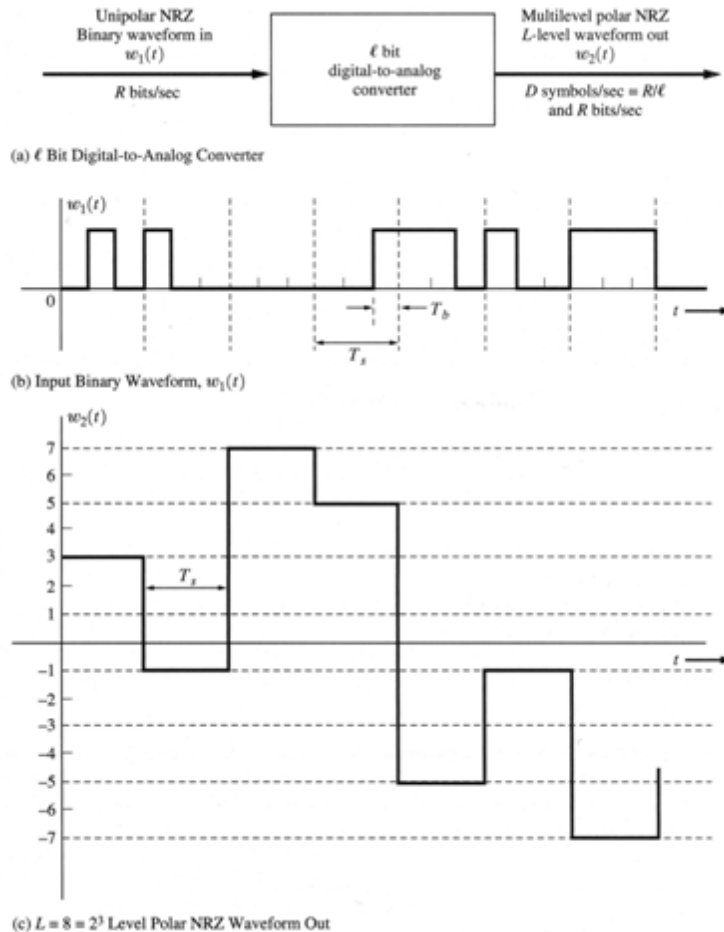


Figure 3-22 Binary-to-multilevel polar NRZ signal conversion.

Multilevel signaling needs less bandwidth than binary signaling because more bits are transmitted per symbol.

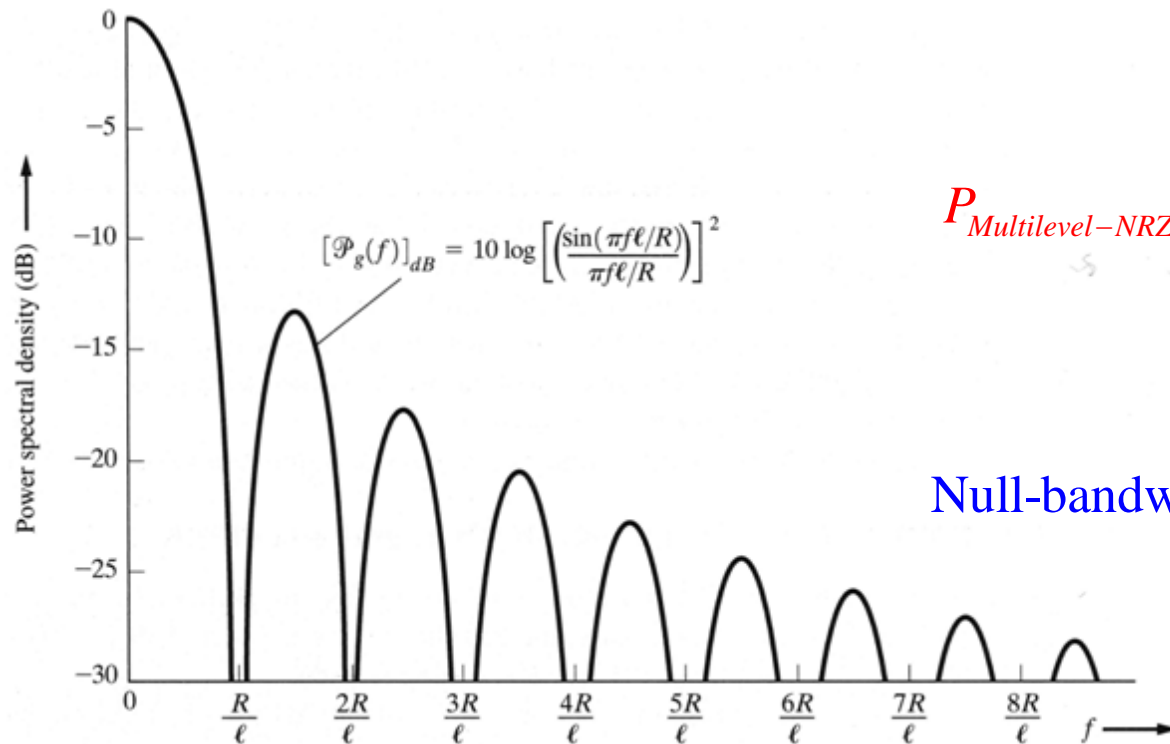
With l bits per symbol:

- transmission levels $\rightarrow L = 2^l$
- symbol rate $\rightarrow D = \frac{R_b}{l}$ [baud]

TABLE 3-5 THREE-BIT DAC CODE

| Digital Word | Output Level, $(a_n)_i$ |
|--------------|-------------------------|
| 000 | +7 |
| 001 | +5 |
| 010 | +3 |
| 011 | +1 |
| 100 | -1 |
| 101 | -3 |
| 110 | -5 |
| 111 | -7 |

Multilevel signaling with rectangular pulses (2)



$$P_{\text{Multilevel-NRZ}}(f) = K \left(\frac{\sin l \pi f T_b}{l \pi f T_b} \right)^2$$

$$= K \text{sinc}^2 l f T_b$$

Null-bandwidth: $B_{\text{null}} = \frac{R}{l}$

For L -level signaling the symbol duration increases to $T_s = l T_b$ with $l = {}^2 \log L$.

Spectral efficiency (1)

Bandwidth is a scarce (expensive) resource!

Spectral efficiency: number of bits/s R_b that can be transmitted in a bandwidth B .

$$\eta \triangleq \frac{\text{bit rate}}{\text{transmission bandwidth}} = \frac{R_b}{B} \quad [(\text{bits/s})/\text{Hz}]$$

With efficient L -level signaling, spectral efficiency is increased:

$$\eta = \frac{LR_s}{B} \quad [(\text{bits/s})/\text{Hz}]$$

Upper bound on spectral efficiency (Shannon):

$$\eta_{\max} \triangleq \frac{C}{B} = \log_2(1+\text{SNR})$$

Spectral efficiency (2)

Design constraint: limit the required bandwidth! \Rightarrow Choose the signaling technique with the **highest spectral efficiency**. Of course you need to consider other boundary conditions as power, bit error probability and cost.

TABLE 3–6 SPECTRAL EFFICIENCIES OF LINE CODES

| Code Type | First Null Bandwidth (Hz) | Spectral Efficiency $\eta = R/B$ [(bits/s)/Hz] |
|----------------------|------------------------------|---------------------------------------------------|
| Unipolar NRZ | R | 1 |
| Polar NRZ | R | 1 |
| Unipolar RZ | $2R$ | $\frac{1}{2}$ |
| Bipolar RZ | R | 1 |
| Manchester NRZ | $2R$ | $\frac{1}{2}$ |
| Multilevel polar NRZ | R/ℓ | ℓ |

Inter-Symbol Interference (1)

When pulses are transmitted in a smaller bandwidth than needed, they will be **distorted**: time dispersion and pulse overlap →

Inter-symbol Interference (ISI).

Question: How can we restrict the channel bandwidth without causing ISI?

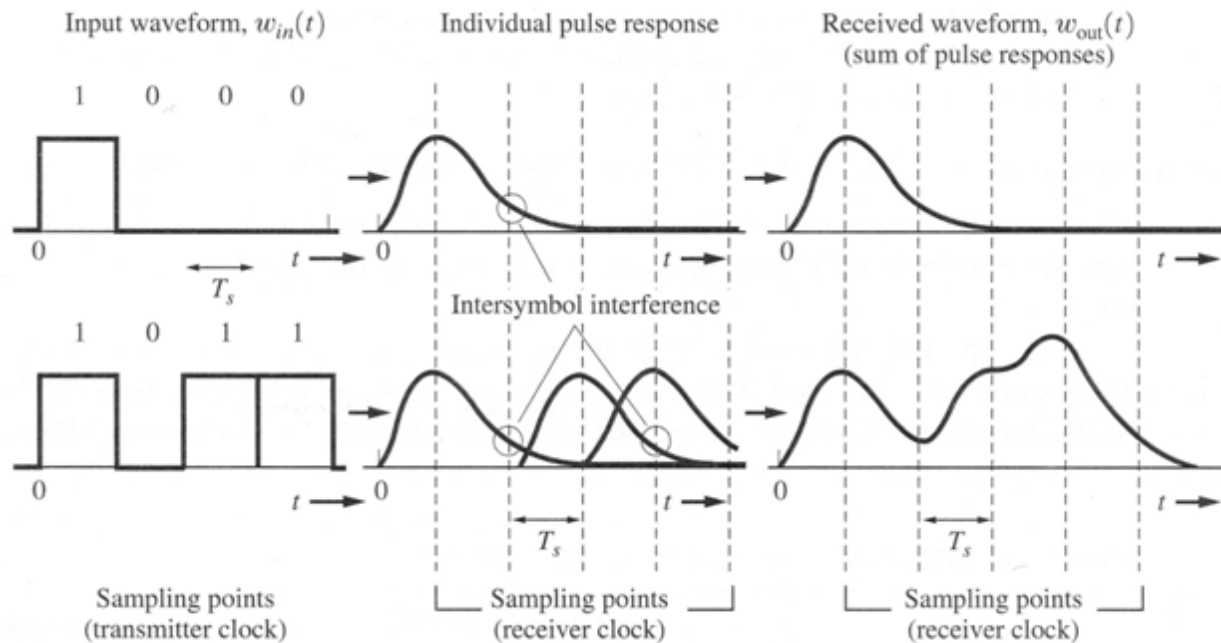


Figure 3-23 Examples of ISI on received pulses in a binary communication system.

Inter-Symbol Interference (2)

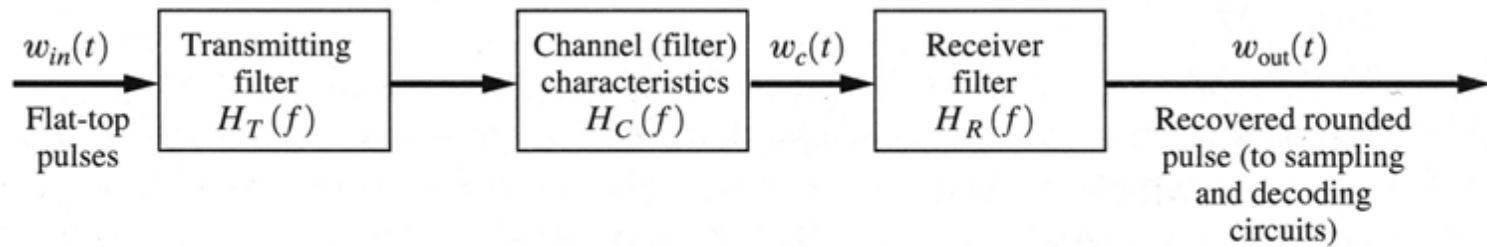


Figure 3–24 Baseband pulse-transmission system.

Let: $w_{in}(t) = \sum_{n=-\infty}^{+\infty} a_n h(t - nT_s)$ with $h(t) = \text{rect}\left(\frac{t}{T_s}\right)$ and symbol rate $D = \frac{1}{T_s} = \frac{1}{lT_b}$.

The values a_n can be those of a multilevel signal.

Now we find for: $w_{in}(t) = \sum_{n=-\infty}^{+\infty} a_n h(t) * \delta(t - nT_s) = \left[\sum_{n=-\infty}^{+\infty} a_n \delta(t - nT_s) \right] * h(t)$

and $w_{out}(t) = \left[\sum_{n=-\infty}^{+\infty} a_n \delta(t - nT_s) \right] * h_e(t)$

with $h_e(t) = h(t) * h_T(t) * h_C(t) * h_R(t)$
 $H_e(f) = H(f) \cdot H_T(f) \cdot H_C(f) \cdot H_R(f)$

Inter-Symbol Interference (3)

For elimination of ISI, the equalizing filter is:

$$H_R(f) = \frac{H_e(f)}{H(f)H_T(f)H_C(f)}$$

Spectrum of the
required pulse shape.

When $H_C(f)$ variable, then $H_R(f)$ should be adaptive (switched telephone line, radio channel).

This requires an adaptive receiver, i.e by means of:

- feedback
- learning filter based on a training-sequence (known word).

Design problem: Determine the optimum $H_T(f)$ and $H_R(f)$ for given $H_C(f)$ and the desired $H_e(f)$.

Elimination of ISI (1)

Nyquist (1928) found three methods to obtain ISI-free transmission.

Nyquist's 1e methode

Requirements for a pulse on the decision circuit's input:

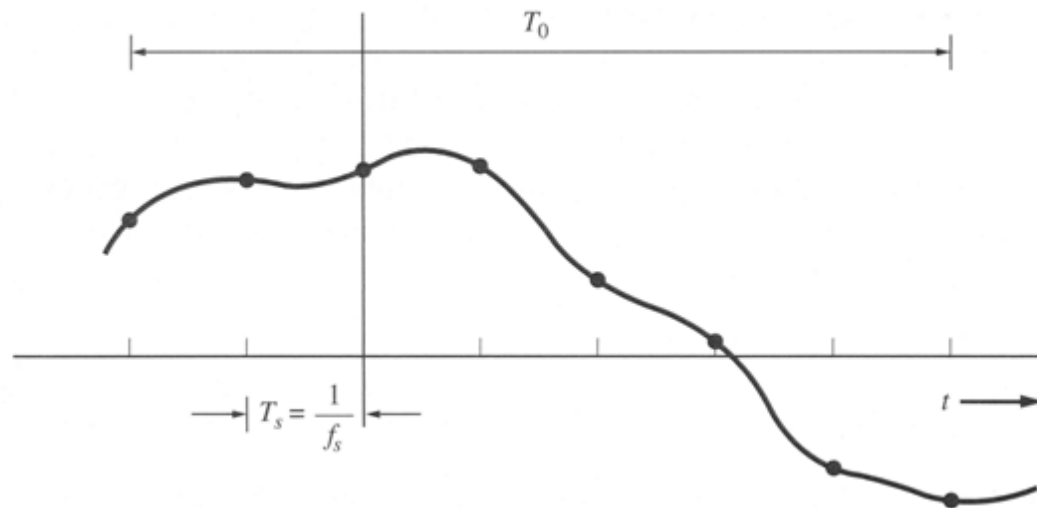
$$h_e(\tau + kT_s) = \begin{cases} C \rightarrow k = 0 \\ 0 \rightarrow k \neq 0 \end{cases}$$

Where τ is a timing offset of the receiver clock. I.e. choose:

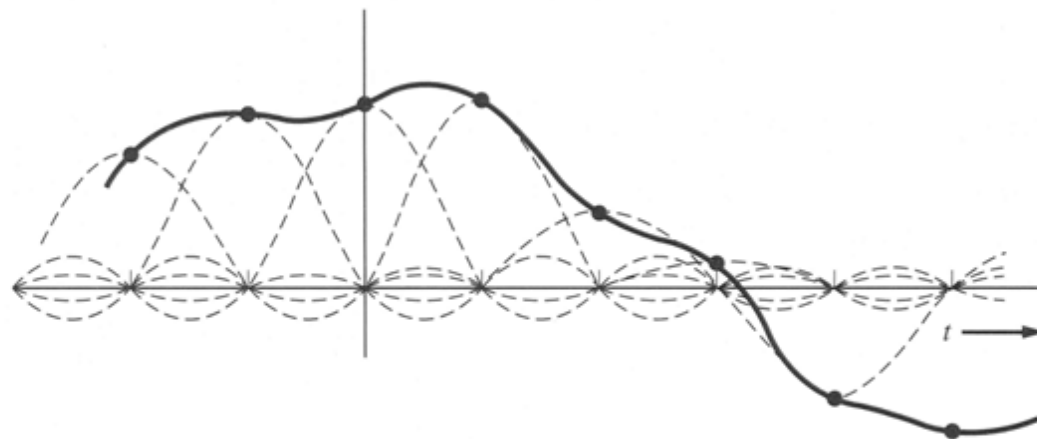
$$h_e(t) = \text{sinc}(f_s t) = \frac{\sin \pi f_s t}{\pi f_s t} \quad \text{for a symbol rate } f_s = \frac{1}{T_s}$$

$$\text{with } H_e(f) = \frac{1}{f_s} \prod \left(\frac{f}{f_s} \right) \Rightarrow B = f_s / 2 ,$$

for which $H_T(f)$ and $H_R(f)$ need to be determined. No ISI will occur since the 1e Nyquist criterion has been fulfilled.



(a) Waveform and Sample Values



(b) Waveform Reconstructed from Sample Values

Figure 2-17 Sampling theorem.

Elimination of ISI (2)

The optimal filter \Rightarrow sinc-pulses:

- minimum bandwidth: $B = f_s / 2$
- symbol rate: $D = 1/T_s = 2B$ [baud]

In practice not realizable since:

- $h_e(t)$ is non-causal: $h_e(t < 0) \neq 0$ and is difficult to implement due to steep skirts in the frequency domain,
- synchronization is very critical; small timing errors result in a large increase of errors due to ISI.

In practice a slightly broader filter is used, i.e. the

Raised Cosine filter

Raised Cosine filter (1)

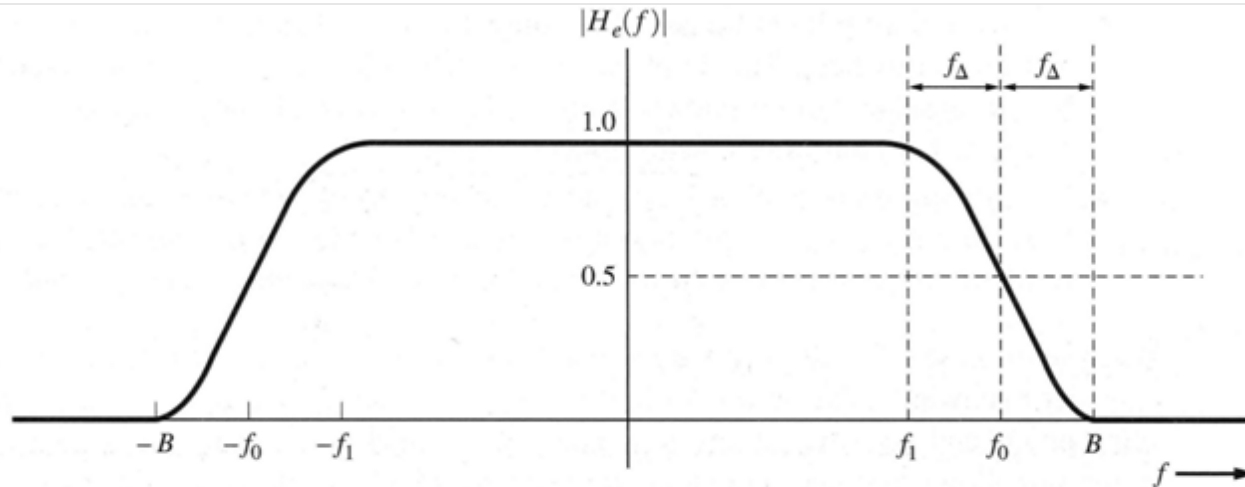


Figure 3-25 Raised cosine-rolloff Nyquist filter characteristics.

With:

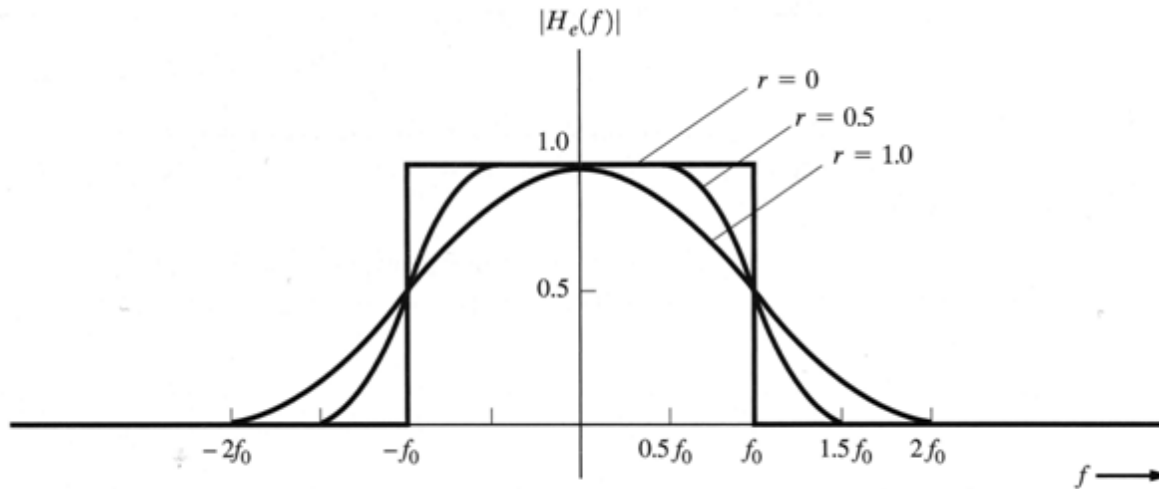
$$f_{\Delta} \triangleq B - f_0 \quad f_0 = \text{-6 dB frequency} = \frac{D}{2}$$

$$f_1 \triangleq f_0 - f_{\Delta}$$

$$\text{Roll-off factor: } r = \frac{f_{\Delta}}{f_0} \quad (0 \leq r \leq 1)$$

$$\text{Absolute bandwidth} = B$$

Raised Cosine filter (2)



(a) Magnitude Frequency Response

$$H_e(f) = \begin{cases} 1 & |f| < f_1 \\ \frac{1}{2} \left(1 + \cos \left(\frac{\pi(|f| - f_1)}{2f_\Delta} \right) \right) & f_1 < |f| < B \\ 0 & |f| > B \end{cases}$$

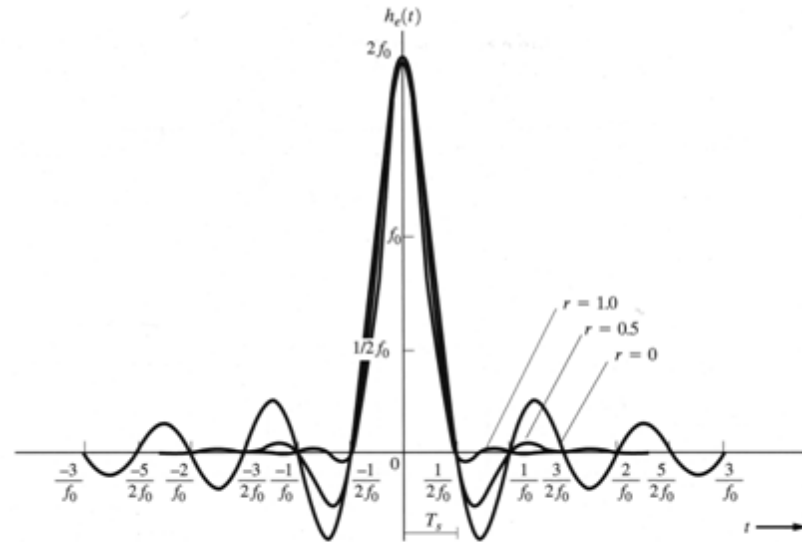
Cosine roll-off of the signal spectrum

Raised Cosine filter (3)

$$h_e(t) = \mathfrak{F}^{-1}\{H_e(f)\}$$

$$= 2f_0 \operatorname{sinc}(2f_0 t) \frac{\cos 2\pi r f_0 t}{1 - (4r f_0 t)^2}$$

A weighted sinc-pulse.



(b) Impulse Response

Figure 3-26 Frequency and time response for different rolloff factors.

- no ISI according to Nyquist_1 for $t = nT_s$ (and $\tau = 0$)
- roll-off factor: $r = \frac{f_\Delta}{f_0} = \frac{B - f_0}{f_0} \Rightarrow r = \frac{B - D/2}{D/2}$
- maximum symbol rate without ISI: $D = 1/T_s = 2f_0 = \frac{2B}{1+r}$
- absolute bandwidth: $B = f_0 + f_\Delta = (1+r)D/2$

Class of ISI-free filters

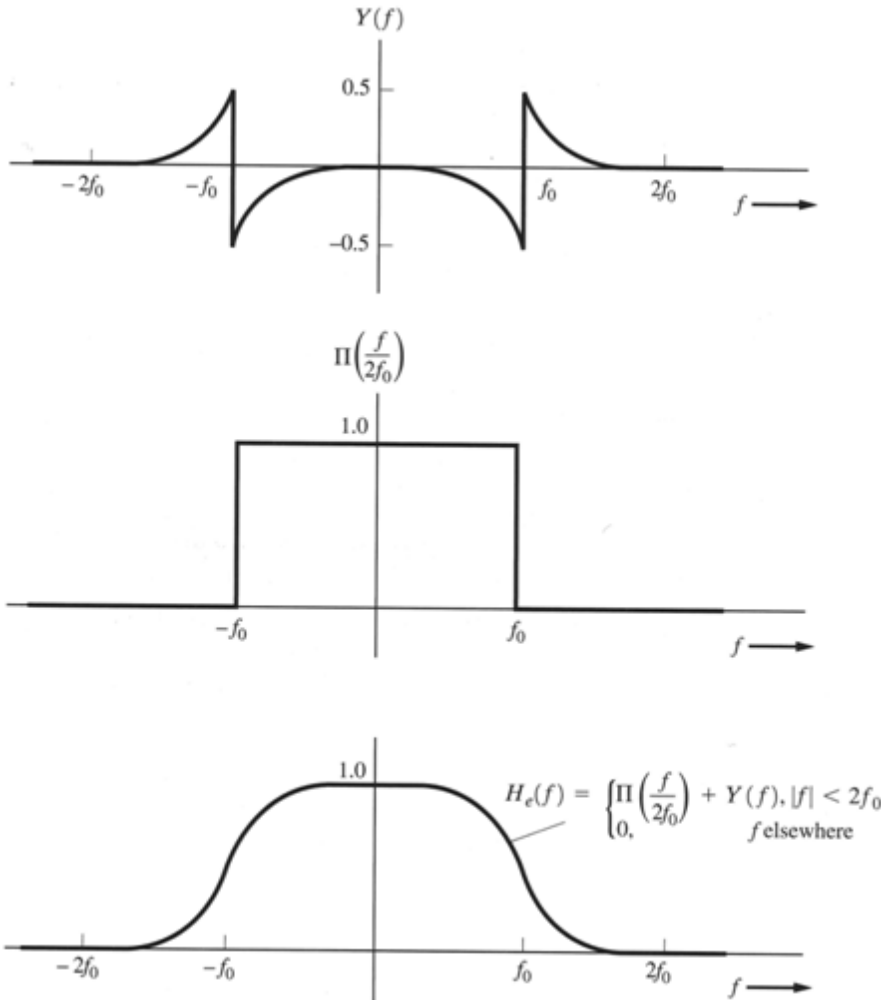


Figure 3-27 Nyquist filter characteristic.

$$H_e(f) = \begin{cases} \Pi\left(\frac{f}{2f_0}\right) + Y(f) & |f| \leq 2f_0 \\ 0 & |f| > 2f_0 \end{cases}$$

With:

1. $Y(f)$ real
2. $Y(f)$ even-symmetric around $f = 0$
3. $Y(f)$ uneven-symmetric $f = f_0$

No ISI for $R_s = 2f_0$

ISI-free filters and noise

With Nyquist_1: ISI = 0 but noise is not minimum!

Using:

$$H_T(f) = \frac{\sqrt{H_e(f)} [P_{n_rx}(f)]^{1/4}}{\alpha |H(f)| \sqrt{H_C(f)}} \quad P_{n_rx}(f) : \text{is noise PSD at the receiver}$$
$$H_R(f) = \frac{\alpha \sqrt{H_e(f)}}{\sqrt{H_C(f)} [P_{n_rx}(f)]^{1/4}}$$

Now we obtain ISI = 0 and minimum noise at the receiver (*matched filter*):

- large transmit power at frequencies with a high noise level at RX,
- receive filter suppresses frequencies with a high noise level

and: $H_e(f) = H(f)H_T(f)H_C(f)H_R(f)$

Other methods of ISI-free filtering

Nyquist's 2e method:

- allow for controlled (known) ISI, which can be compensated in a later stage.

Nyquist's 3rd method:

- not based on $h_e(nT)$ but on $\int h_e(t)dt$, i.e. the area under the equivalent impulse.