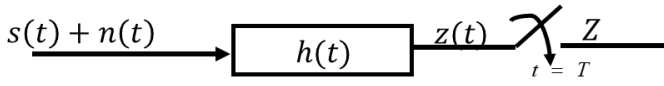


- Matched Filter System Model



information signal pulse $s(t)$

background (AWGN) noise $n(t)$

filter impulse response $h(t)$

$$z(t) = s_f(t) + n_f(t)$$

where

$$s_f(t) = h(t) * s(t)$$

$$n_f(t) = h(t) * n(t)$$

Decision variable $Z = z(t = T) = s_f(T) + n_f(T)$

Design Goal: Find $h(t)$ which maximizes the signal to noise given as

$$\gamma = \frac{|s_f(T)|^2}{E[|n_f(T)|^2]}$$

- Signal to Noise Ratio (SNR)

$$s_f(T) = h(t) * s(t)|_{t=T} = \int_{-\infty}^{\infty} H(f)S(f)e^{j2\pi sTf} df \Big|_{t=T}$$

$$= \int_{-\infty}^{\infty} H(f)S(f)e^{j2\pi sTf} df$$

$$E[|n_f(T)|^2] = \text{power of } n_f(t) \because n \text{ is ergodic}$$

$$= \text{total area of PSD of } n_f(t)$$

$$\text{PSD of } n_f(t) = |H(f)|^2 \cdot \text{PSD of } n(t) = \frac{N_0}{2} |H(f)|^2$$

$$\therefore E[|n_f(T)|^2] = \frac{N_0}{2} \int_{-\infty}^{\infty} |H(f)|^2 df$$

$$\gamma = \frac{|s_f(T)|^2}{E[|n_f(T)|^2]} = \frac{|\int_{-\infty}^{\infty} H(f)S(f)e^{j2\pi sTf} df|^2}{\frac{N_0}{2} \int_{-\infty}^{\infty} |H(f)|^2 df}$$

- Find Maximum of SNR

$$H_{\max \gamma} = \underset{H(f)}{\operatorname{argmax}} \gamma$$

$$= \underset{H(f)}{\operatorname{argmax}} \frac{|\int_{-\infty}^{\infty} H(f)S(f)e^{j2\pi sTf} df|^2}{\frac{N_0}{2} \int_{-\infty}^{\infty} |H(f)|^2 df}$$

$$\int_{-\infty}^{\infty} x(t)y^*(t) dt \text{ is maximized when } x(t)$$

$$= ky(t) \text{ where } k = \text{Natural Number}$$

$$\therefore \int_{-\infty}^{\infty} H(f)S(f)e^{j2\pi sTf} df \text{ is maximized when}$$

$$H(f) = k[S(f)e^{j2\pi sTf}]^* = kS(f)^*e^{-j2\pi sTf}$$

- Matched Filter

$$H_{\max \gamma}(f) = kS(f)^*e^{-j2\pi sTf} \xrightarrow{\mathcal{F}^{-1}} h_{\max \gamma}(t) = kS^*(T - t)$$

The filter with an impulse response $h_{\max \gamma}(t)$ maximizes the signal to noise ratio. This filter is called 'Matched Filter'.

- NSR of Matched Filter

$$\gamma_{\max} = \frac{|\int_{-\infty}^{\infty} H(f)S(f)e^{j2\pi sTf} df|^2}{\frac{N_0}{2} \int_{-\infty}^{\infty} |H(f)|^2 df}$$

$$= \frac{|\int_{-\infty}^{\infty} kS(f)^*e^{-j2\pi sTf} S(f)e^{j2\pi sTf} df|^2}{\frac{N_0}{2} \int_{-\infty}^{\infty} |S(f)|^2 df}$$

$$= \frac{|\int_{-\infty}^{\infty} k|S(f)|^2 df|^2}{\frac{N_0}{2} \int_{-\infty}^{\infty} |S(f)|^2 df}$$

$$= \frac{\frac{N_0}{2} \int_{-\infty}^{\infty} |kS(f)|^2 df}{\frac{N_0}{2} \int_{-\infty}^{\infty} |S(f)|^2 df} = \frac{\int_{-\infty}^{\infty} |S(f)|^2 df}{\int_{-\infty}^{\infty} |S(f)|^2 df} = \frac{2E_s}{2} = E_s$$

$$\because \int_{-\infty}^{\infty} |S(f)|^2 df = \int_{-\infty}^{\infty} |s(t)|^2 dt = E_s \text{ (Parseval theorem)}$$

The matched filter is determined by the employed pulse shape. However, the performance (output SNR) of matched filter does not depend on the employed pulse shape. It depends only on the employed Energy.

- Pulse Shaping Process

Consider a system that transmit the consecutive data pulses with a pulse $h_t(t)$.

$$x(t) = \sum_{n=0}^{\infty} x_n h_t(t - nT)$$

where x_n are data symbol sequences and T is symbol duration.

- Delta(Impulse) Modulation

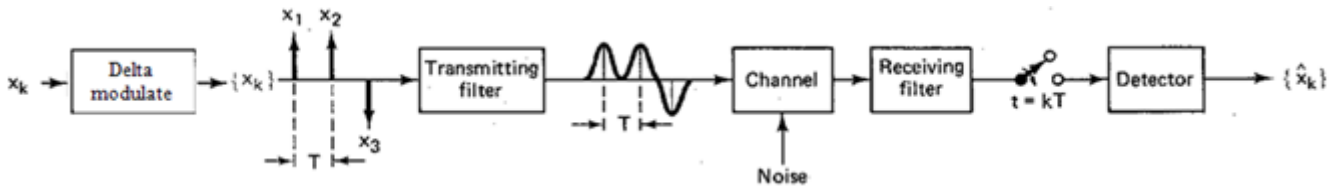
$$x(t) = \sum_{n=0}^{\infty} x_n h_t(t - nT)$$

$$= \sum_{n=0}^{\infty} x_n h_t(t) * \delta(t - nT)$$

$$= h_t(t) * \sum_{n=0}^{\infty} x_n \delta(t - nT)$$

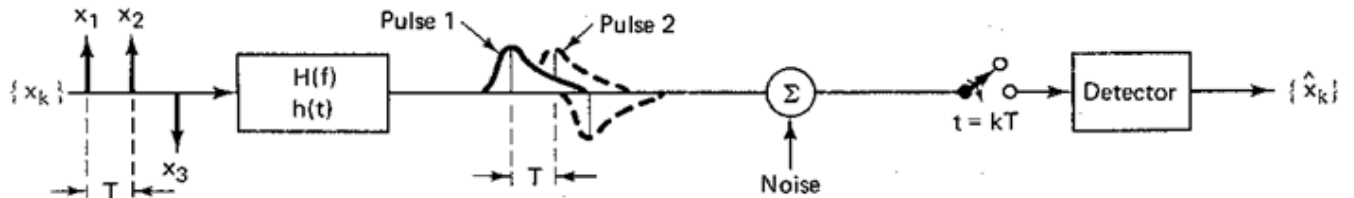
$$= h_t(t) * \text{delta modulation of } x_n$$

- Overall System: Tx, Channel and RX



Impulse response $h_t(t)$ from Transmitting filter and $h_c(t)$ from Channel and $h_r(t)$ from Receiving filter.

Therefore, equivalent system is



where $h(t)$ is $h_t(t) * h_c(t) * h_r(t)$.

- Inter-Symbol Interference (ISI)

The pulses are overlapped and thus, the detected symbol sequence is

$$\hat{x}_k = x_k + n_k + \sum_{i \neq k} \alpha_i x_i$$

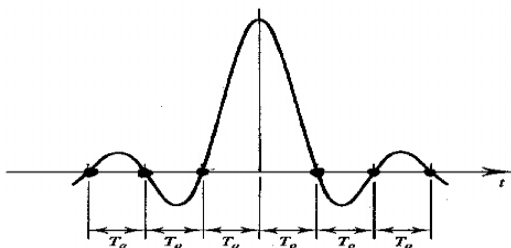
where the term $\sum_{i \neq k} \alpha_i x_i$ is called ISI.

- Pulse Shape Design

Use a very short pulse to avoid ISI but large bandwidth (BW).

Pulse shape design issues are no ISI and BW as small as possible

Therefore, ISI free condition is $h(zT)$ (where $z \in \mathbb{Z} - \{0\}$) are all 0, i.e. $h(t)$ crosses 0 where t is all integer (except 0) multiples of symbol duration T .



- Mathematical Setup for ISI Free Condition.

$$h(t) \xrightarrow{t=kT} \delta(k)$$

where $h(t)$ is ISI free pulse.

Then the spectrum of sampled $h(t)$ is

$$\sum H(f + zR_s)$$

where $R_s = 1/T$, i.e., symbol rate.

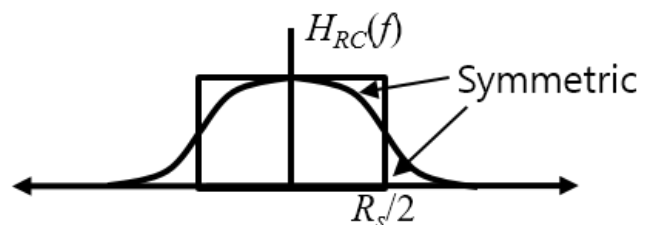
Consequently, ISI free condition in frequency domain $\sum H(f + zR_s)$ should be equal to spectrum of impulse, i.e., a constant.

- Raised Cosine (RC) Pulse

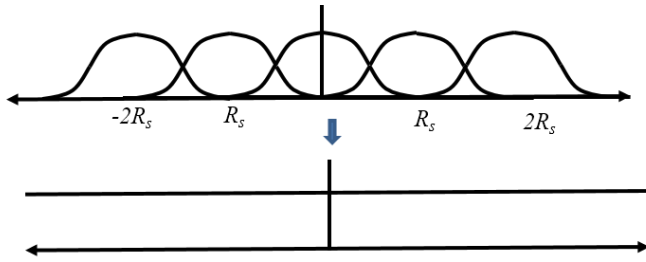
$$H_{RC}(f) = \begin{cases} 1, & \text{for } |f| < W_0 - W \\ \cos^2\left(\frac{\pi}{4} \cdot \frac{|f| + W - 2W_0}{W - W_0}\right), & \text{for } W_0 - W < |f| < W \\ 0, & \text{for } |f| > W \end{cases}$$

where

$$W_0 = \frac{1}{2T} = \frac{R_s}{2} [\text{Hz}] \text{ Nyquist minimum bandwidth}$$



- Spectrum of RC Pulse



RC pulse has a constant spectrum after sampling. So, RC pulse is most commonly used ISI free pulse.

- RC Pulse Spectrums with Different Bandwidths

Minimum BW $W_{\min} = 0.5R_s$ and

Excessive BW $W_{\text{EX}} = W - W_{\min} = W - 0.5R_s$

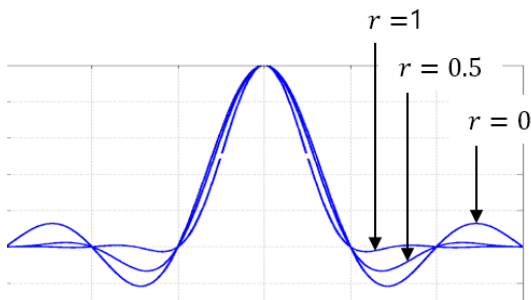
Roll off factor

$$r \triangleq \frac{W_{\text{EX}}}{W_{\min}} = \frac{W - 0.5R_s}{0.5R_s}$$

where W is Zero to Null Bandwidth.

Note that $H_{RC}(\pm 0.5R_s) = 0.5$ irrespective of the roll off factor that is 3dB BW which is 6dB in power spectrum.

- BW vs. Truncation Error Tradeoff



- Small BW, Small r , Large ripples, Large error.
- Large BW, Large r , Small ripples, Small error

- Square Root Raised Cosine Pulse

$$h(t) = h_t(t) * h_c(t) * h_r(t) \\ \rightarrow H(f) = H_t(f)H_c(f)H_r(f)$$

Assume the ideal channel, i.e.,

$$h_c(t) = \delta(t) \rightarrow H_c(f) = 1$$

Then, to maximize output SNR of the receiver filter, $h_t(t)$ should be the matched filter to the TX pulse $h_r(t)$, i.e.,

$$h_t(t) = h_r^*(-t) \rightarrow H_r(f) = H_t^*(f)$$

Substituting,

$$H(f) = H_t(f)H_c(f)H_r(f) = |H_t(f)|^2$$

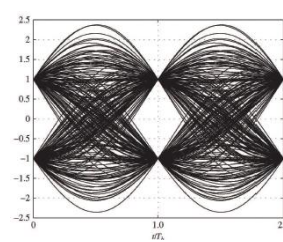
Note that we design $h_t(t)$, however, $h(t)$ not $h_t(t)$ should be ISI-free pulse. Hence, $|H_t(f)|^2$ should be ISI-free spectrum.

$$|H_t(f)|^2 = H_{RC}(f) \rightarrow H_t(f) = \sqrt{H_{RC}(f)}$$

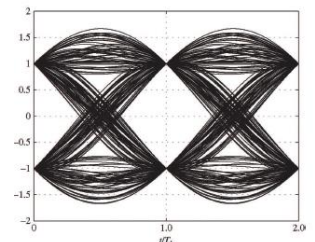
where $H_{RC}(f)$ is called square root raised cosine (SRRC) pulse and is denoted by $H_{SRRC}(f)$

- Eye Diagram

Partitioned and overlaid waveforms. Partition length is integer multiples of the symbol duration. Easy to see ISI and visually assess the signaling performance.



Small roll-off factor:
narrow eye opening



Large roll-off factor:
wide eye opening

- Why and When the Waveform-level Simulation Needed?

- Accommodate the waveform-level effects such as:

Channel effects (multipath fading, clipping, etc.) / Synchronization error (phase, frequency, symbol timing)

- Accommodate the waveform-assisted (or waveform-level) algorithms such as:

Pulse shaping and matched filtering / Synchronization / Equalization

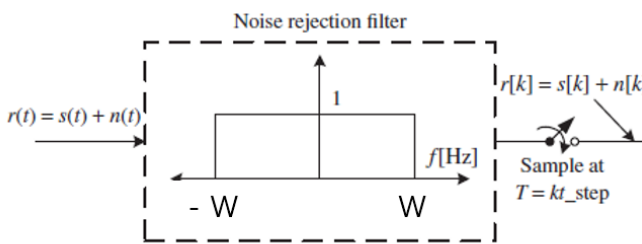
- Observe the waveforms

Eye diagram / Constellation / Signal trajectory / Peak to average power ratio (PAPR)

- Limitations of waveform-level simulation in the computers.

- Non-ideally continuous waveforms, i.e., we have to use the sampled versions of the waveforms
- Careful setting needed for the sampling interval, variance for the sampled version of the noise signal.

- Noise Rejection Filter



To pass the signal $s(t)$ as is and to reject white noise.

To make the output noise sequence $n[k]$ be a white Gaussian process, the sampling frequency should be equal to $2W$

$$\frac{1}{t_{\text{step}}} = 2W \rightarrow \frac{1}{2 \cdot t_{\text{step}}} = W$$

- Quaternary Phase Shift Keying (QPSK)

$$s_i(t) = \sqrt{\frac{2E}{T}} \cos(\omega_0 t - \phi_i), \quad 0 \leq t \leq T, i = 1, 2, 3, 4$$

$$= \sqrt{E} \cos(\phi_i) \sqrt{\frac{2}{T}} \cos(\omega_0 t) + \sqrt{E} \sin(\phi_i) \sqrt{\frac{2}{T}} \sin(\omega_0 t)$$

$$\psi_1(t) = \sqrt{\frac{2}{T}} \cos(\omega_0 t), \quad \psi_2(t) = \sqrt{\frac{2}{T}} \sin \omega_0 t$$

then,

$$s_i(t) = \sqrt{E} \cos(\phi_i) \sqrt{\frac{2}{T}} \psi_1(t) + \sqrt{E} \sin(\phi_i) \sqrt{\frac{2}{T}} \psi_2(t) = x(t) + y(t)$$

where

$$x(t) = \begin{cases} \sqrt{E_b} \psi_1(t), & \text{if } b_I = '1' \\ -\sqrt{E_b} \psi_1(t), & \text{if } b_I = '0' \end{cases}$$

$$y(t) = \begin{cases} \sqrt{E_b} \psi_2(t), & \text{if } b_Q = '1' \\ -\sqrt{E_b} \psi_2(t), & \text{if } b_Q = '0' \end{cases}$$

and $x(t)$ and $y(t)$ are BPSK signals. So, QPSK is the summation of two parallel BPSK signals that are orthogonal each other.

It is intuitive that BPSK and QPSK have the identical BER. With identical bandwidth, QPSK has twice data rate than BPSK.

4-ary symbol index i	$b_I b_Q$	ϕ_i
1	11	$\pi/4$
2	10	$-\pi/4$
3	01	$3\pi/4$
4	00	$-3\pi/4$

- Pulse Shaped QPSK

$$s(t) = I(t) \sqrt{\frac{2}{T_s}} \cos(\omega_c t) + Q(t) \sqrt{\frac{2}{T_s}} \sin(\omega_c t)$$

$$= \sqrt{\frac{2}{T_s} (I^2(t) + Q^2(t))} \cos\left(\omega_c t - \tan^{-1}\left(\frac{Q(t)}{I(t)}\right)\right)$$

- Offset QPSK (OQPSK)

To avoid the simultaneous polarity changes of $I(t)$, and $Q(t)$, simply insert a delay block into one of $I(t)$ and $Q(t)$ signal before multiplying the sinusoids.

- MPSK Signal Expression

$$s_i(t) = \sqrt{\frac{2E_s}{T}} \cos\left(\omega_c t - \frac{2\pi(i-1)}{M}\right)$$

$$0 \leq t \leq T, i = 1, 2, \dots, M$$

$$s_i(t) = \sqrt{E_s} \cos\left(\frac{2\pi(i-1)}{M}\right) \sqrt{\frac{2}{T}} \cos(\omega_c t) + \sqrt{E_s} \sin\left(\frac{2\pi(i-1)}{M}\right) \sqrt{\frac{2}{T}} \sin(\omega_c t)$$

$$= a_{i1}\psi_1(t) + a_{i2}\psi_2(t)$$

where

$$\psi_1(t) = \sqrt{\frac{2}{T}} \cos(\omega_c t), \quad a_{i1} = \sqrt{E_s} \cos\left(\frac{2\pi(i-1)}{M}\right)$$

$$\psi_2(t) = \sqrt{\frac{2}{T}} \sin(\omega_c t), \quad a_{i2} = \sqrt{E_s} \sin\left(\frac{2\pi(i-1)}{M}\right)$$

thus, in vector space,

$$s_i(t) \rightarrow s_i = (a_{i1}, a_{i2})$$

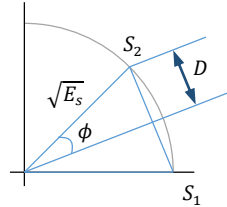
$$= \left(\sqrt{E_s} \cos\left(\frac{2\pi(i-1)}{M}\right), \sqrt{E_s} \sin\left(\frac{2\pi(i-1)}{M}\right) \right)$$

- Symbol Error of MPSK

$$\phi = \frac{1}{2} \left(\frac{2\pi}{M} \right) = \frac{\pi}{M}, D = \sqrt{E_s} \sin \phi$$

Minimum distance energy

$$E_d = (2D)^2 = (2\sqrt{E_s} \sin \phi)^2$$



where $E_s = \log_2 M \cdot E_b$

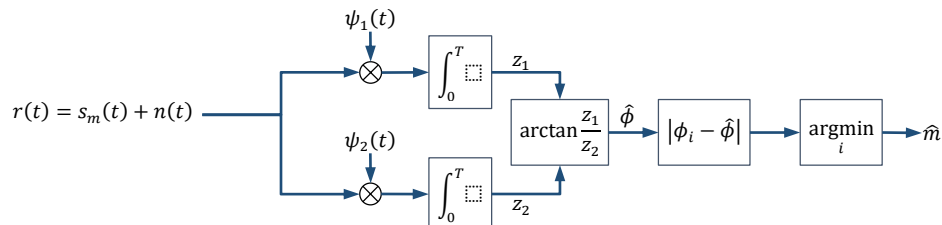
$$\therefore E_d = \left(2\sqrt{\log_2 M \cdot E_b} \sin \frac{\pi}{M} \right)^2$$

$$= 4 \log_2 M \cdot E_b \sin^2 \frac{\pi}{M} \stackrel{\text{for large } M}{\approx} 4 \log_2 M \cdot E_b \left(\frac{\pi}{M} \right)^2$$

$$= 4E_b \pi^2 \frac{\log_2 M}{M^2}$$

As M increases, E_d decrease. Therefore, the symbol error rate increases.

- MPSK Demodulator



- Gray mapping

Neighboring symbols have only 1-bit difference. This reduce bit errors.

- BER of MPSK

For large SNR, the errors between the closest symbol pairs in vector space is dominant. So,

$$P_s \approx Q\left(\frac{D}{\sqrt{N_0/2}}\right) = Q\left(\frac{\sqrt{E_D}/2}{\sqrt{N_0/2}}\right) = Q\left(\sqrt{\frac{E_D}{2N_0}}\right)$$

$$\text{BER of } P_s \approx \frac{1}{\log_2 M} P_s = \frac{1}{\log_2 M} Q\left(\sqrt{\frac{E_D}{2N_0}}\right)$$

$$\text{where } E_d \approx 4E_b \pi^2 \frac{\log_2 M}{M^2}$$

There are three assumptions: large SNR, gray-mapped symbols and large M .

As M increase, BER of MPSK increases.

- Bandwidth of MPSK

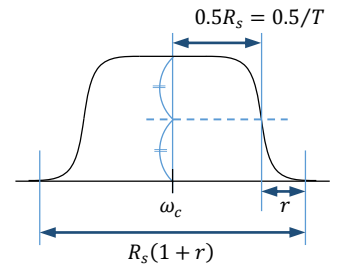
$$W = R_s(1 + r)$$

Where

$$R_s = \frac{1}{T} = \frac{1}{\log_2 M \cdot T_b}$$

$$= \frac{R_b}{\log_2 M}$$

r = roll-off factor, T_b = bit duration, R_b = bit rate



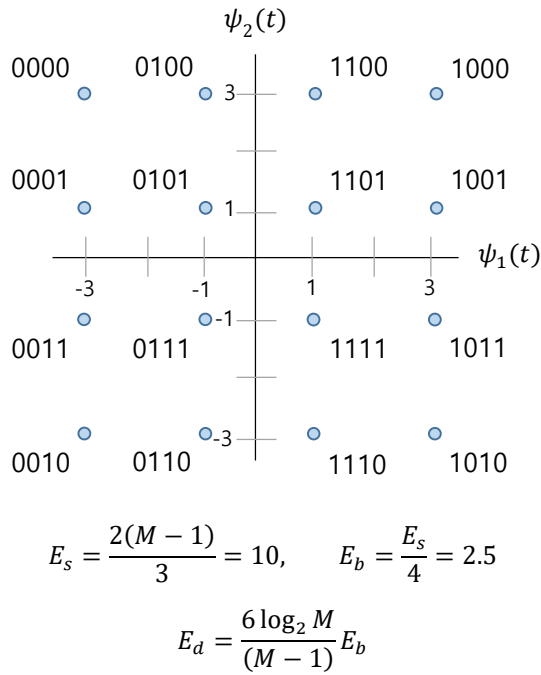
thus,

$$W = \frac{R_b(1 + r)}{\log_2 M}$$

As M increases with a fixed bit rate R_b , the bandwidth of MPSK decreases.

As M increases with a fixed bandwidth, the bit rate increases.

- 16QAM Signal



- 16QAM Demodulator

$$\hat{b}_4 = \begin{cases} 1, & z_1 > 0 \\ 0, & \text{else} \end{cases}, \quad \hat{b}_3 = \begin{cases} 1, & |z_1| < 2 \\ 0, & \text{else} \end{cases}$$

$$\hat{b}_2 = \begin{cases} 1, & z_2 > 0 \\ 0, & \text{else} \end{cases}, \quad \hat{b}_1 = \begin{cases} 1, & |z_2| < 2 \\ 0, & \text{else} \end{cases}$$

- Bandwidth of QAM as same as MSPK

- Bandwidth Efficiency

$$\text{BW Efficiency} = \frac{\text{Bit rate}}{\text{Bandwidth}} = \frac{R_b}{W} = \frac{\log_2 M}{1 + r}$$

- MFSK Signal

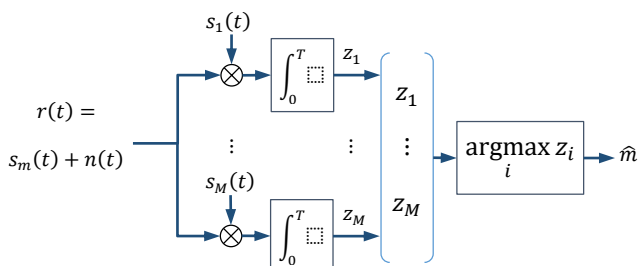
$$s_i(t) = \sqrt{\frac{2E_s}{T}} \cos(\omega_i t) = \sqrt{\frac{2E_s}{T}} \cos(\omega_c t + (i-1)\Delta\omega t)$$

$$\Delta f = \frac{\Delta\omega}{2\pi} = \frac{1}{2T} \quad \text{for coherent MFSK}$$

$$\Delta f = \frac{1}{T} \quad \text{for non-coherent MFSK}$$

$$\langle s_i(t), s_j(t) \rangle = \int_0^T s_i(t) s_j(t) dt = 0, \quad \text{for } \forall i \neq j$$

- MFSK Demodulator: Coherent

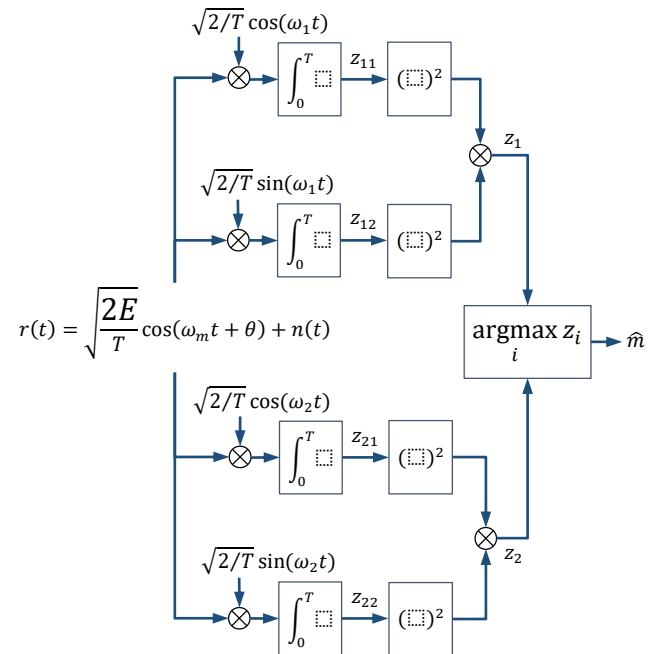


- MFSK Demodulator: Non-coherent

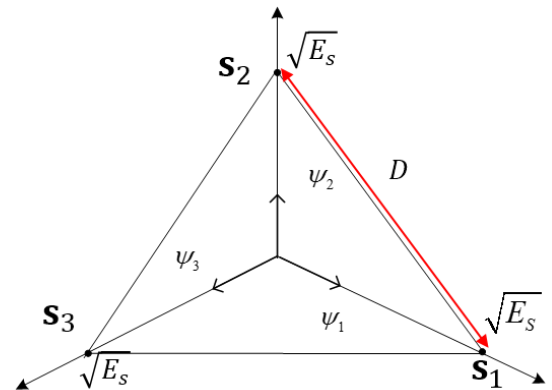
This case when the phase of the local carrier is not synchronized to the received signal.

Mathematically equivalent to set

$$r(t) = \sqrt{\frac{2E}{T}} \cos(\omega_m t + \theta) + n(t) \quad \text{where } \theta = \text{phase error}$$



- Error performance and BW of MFSK



$$E_d = D^2 = 2E_s = 2 \log_2 M \cdot E_b$$

Note that as M increases, E_d increases. Therefore, better BER performance.

- Bandwidth of MFSK

$$W \cdot \Delta f = \begin{cases} \frac{M}{2T} = \frac{M}{2T_b \log_2 M} = \frac{MR_b}{2 \log_2 M}, & \text{coherent} \\ \frac{MR_b}{\log_2 M}, & \text{non-coherent} \end{cases}$$

Note that as M increases, W increases.

- Rayleigh Fading

$$r = hs + n$$

where s is complex plane representation for 2D modulated signal.

$$s = s_1 + js_2, \quad E_s = E[|s|^2]$$

where n is complex Gaussian noise.

$$n = n_1 + jn_2, \quad n_1, n_2 \sim N(0, \frac{N_0}{2})$$

n_1 and n_2 are independent each other.

where h is signal scaling coefficient due to fading.

$$h = z + jy, \quad z, y \sim N(0, \frac{1}{2})$$

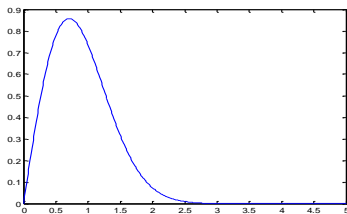
z and y are i.i.d. Gaussian.

- Rayleigh Fading Decision Variable

$$\begin{aligned} D &= re^{-j\angle h} = (hs + n)e^{-j\angle h} = (|h|e^{j\angle h}s + n)e^{-j\angle h} \\ &= |h|s + ne^{-j\angle h} \end{aligned}$$

where signal term $|h|s$ has the same phase as that of the original symbol s .

- PDF of $|h|$



$$f_{|h|}(x) = \begin{cases} 2xe^{-x^2}, & x \geq 0 \\ 0, & x < 0 \end{cases}$$

This is called Rayleigh distribution.

- Instantaneous Symbol Energy

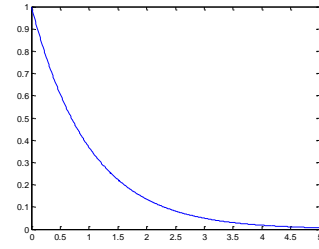
$$|h|^2 E_s$$

where $|h|^2$ is instantaneous symbol energy scaling term. Denote this scaling term as a new RV $c = |h|^2$.

Under Rayleigh fading, instantaneous symbol energy is given by

$$E_i = |h|^2 E_s$$

- PDF of c



$$f_c(x) = \begin{cases} e^{-x}, & x \geq 0 \\ 0, & x < 0 \end{cases}$$

This is called Exponential distribution

- Average Instantaneous Symbol Energy

$$\bar{E}_i = E[|h|^2] \cdot E_s = E[c] \cdot E_s$$

where

$$E[c] = \int_{-\infty}^{\infty} x f_c(x) dx = \int_{-\infty}^{\infty} x e^{-x} dx = 1$$

So, $\bar{E}_i = E_s$

- Average BER under Rayleigh Fading

$$\begin{aligned} p_b &= E_c \left[Q \left(\sqrt{\frac{2cE_b}{N_0}} \right) \right] = \int_{-\infty}^{\infty} Q \left(\sqrt{\frac{2xE_b}{N_0}} \right) f_c(x) dx \\ &= \int_0^{\infty} \left(\int_{\frac{2xE_b}{N_0}}^{\infty} \frac{e^{-t^2/2}}{\sqrt{2\pi}} dt \right) e^{-x} dx \\ &= \frac{1}{2} \left(1 - \sqrt{\frac{E_b/N_0}{1 + E_b/N_0}} \right) \end{aligned}$$

```
clear
syms c EbN0 positive
instantBER = 0.5 * (1-erf(sqrt(c * EbN0)));
BER_fading = int(instantBER * exp(-c), c, 0, inf);
Pretty((BER_fading))
```

```
>>
1      sqrt(EbN0)
- - - - -
2      2sqrt(EbN0 + 1)
```

if $E_b/N_0 \rightarrow 1$

$$p_b = \frac{1}{2} \left(1 - \sqrt{\frac{E_b/N_0}{1 + E_b/N_0}} \right) \approx \frac{1}{4 E_b/N_0}$$

in log scale,

$$\begin{aligned} p_b &= \log\left(\frac{1}{4\gamma}\right) = \log\left(\frac{1}{4}\right) - \log \gamma \\ &= \log\left(\frac{1}{4}\right) - 0.1(10 \log E_b/N_0) \\ &= -6 - 0.1(E_b/N_0 [\text{dB}]) \end{aligned}$$

- **Selection Diversity Combining (SDC)**

$$k_{\text{best}} = \underset{k}{\operatorname{argmax}} |h(k)|$$

$$D = \gamma(k_{\text{best}}) e^{-j\angle(k_{\text{best}})}$$

The branch with the largest fading coefficient is selected for detection and the rest are not used.

When $|h(1)|, |h(2)|, \dots, |h(L)|$ are similar, the considerable symbol energies distributed in the unselected branches are missed.

- **Equal Gain Combining (EGC)**

$$D = \sum_{k=1}^L \gamma(k) e^{-j\angle h(k)}$$

$$= |h(1)|s + n'(1) + |h(2)|s + n'(2) + \dots$$

$$+ |h(k)|s + n'(k)$$

$$= (|h(1)| + |h(2)| + \dots + |h(k)|)s$$

$$+ (n'(1) + n'(2) + \dots + n'(k))$$

The term prior to performing summation is required so that the fading coefficients are coherently summed.

The relatively noisy (=relatively small h) branches contribute rather destructively to the overall SNR of D .

- **Maximum Ratio Combining (MRC)**

$$D = \sum_{k=1}^L g(k) \gamma(k) e^{-j\angle h(k)}$$

$$\rightarrow \sum_{k=1}^L g_{\text{opt}}(k) \gamma(k) e^{-j\angle h(k)}$$

$$= \sum_{k=1}^L \alpha |h(k)| \gamma(k) e^{-j\angle h(k)}$$

$$= \alpha \sum_{k=1}^L |h(k)| e^{-j\angle h(k)} \gamma(k)$$

$$= \alpha \sum_{k=1}^L h^*(k) \gamma(k)$$

Large $g(k)$ to the reliable branches.

Small $g(k)$ to the noisy branches.

We can show that the optimal gains are proportional to the channel fading magnitude.

$$g_{\text{opt}}(k) = \alpha |h(k)|$$

where α is any arbitrary positive real number.

- **Optimal Combining Gain**

$$D = \sum_{k=1}^L g(k) e^{-j\angle h(k)} \gamma(k)$$

$$= \sum_{k=1}^L g(k) e^{-j\angle h(k)} (h(k)s + n(k))$$

$$= \sum_{k=1}^L g(k) |h(k)| s + \sum_{k=1}^L g(k) n(k) e^{-j\angle h(k)} s$$

$$\text{SNR} = \frac{\text{Signal term}^2}{\text{Variance of noise term}} = \frac{(\sum_{k=1}^L g(k) |h(k)| s)^2}{\frac{N_0}{2} \sum_{k=1}^L g^2(k)}$$

$$= \frac{2s^2}{N_0} \cdot \frac{|\langle G, H \rangle|^2}{|G|^2}$$

Inner product is maximized when two vectors have the same angle.

$$G = \alpha H$$

-