

## Disclaimer

This summary is part of the lecture “ETH Communication Systems” (227-0121-00) by Prof. Dr. Armin Wittneben (FS19). It is based on the lecture.

Please report errors to [huettern@student.ethz.ch](mailto:huettern@student.ethz.ch) such that others can benefit as well.

The upstream repository can be found at <https://github.com/noah95/formulasheets>

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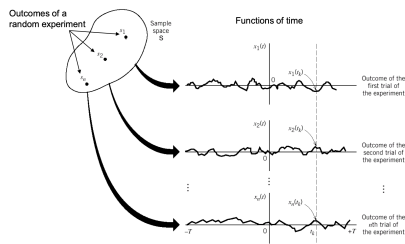
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# ETH Communication Systems 2019

Noah Huetter

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## 1 Random Processes



A random process  $X(t)$ :

- is a sample space composed of (real valued) time functions:  $\{x_1(t), x_2(t), \dots, x_n(t)\}$
- observed at a fixed  $t_k$  is a random variable  $X(t_k) = \{x_1(t_k), x_2(t_k), \dots, x_n(t_k)\}$
- The time function  $x_s(t)$  is a **realization** (sample function)
- $x_s(t_k)$  observed at  $t_k$  is a real number
- A stochastic process consists of infinitely many random variables, one for each  $t_k$ , with the CDF  $F_{\{X(t_k)\}}(x) = P(X(t_k) \leq x)$

### 1.1 Stationary processes

A process is **Strict Sense Stationary (SSS)** if:

- $X(t)$  and  $X(t+\tau)$  have same statistics  $\forall \tau$
- The joint distribution function of a set of r.v. observed at times  $t_1, \dots, t_n$  is invariant to a time-shift.

$$\forall n, \tau, t_1, \dots, t_n : \\ F_{\{X(t_1+\tau), X(t_2+\tau), \dots, X(t_n+\tau)\}}(x_1, x_2, \dots, x_n) = \\ F_{\{X(t_1), X(t_2), \dots, X(t_n)\}}(x_1, x_2, \dots, x_n)$$

Properties:

$$\begin{aligned} \forall t_k : \mu_X(t_k) &= \mu_X \\ \forall t_1, t_2 : R_X(t_1, t_2) &= R_X(t_2 - t_1) = R_X(\tau) \\ C_X(t_1, t_2) &= E[(X(t_1) - \mu_X)(X(t_2) - \mu_X)] \\ &= R_X(t_2 - t_1) - \mu_X^2 \end{aligned}$$

A process is **Wide Sense Stationary (WSS)** if a r.p. has a *constant* mean and the autocorrelation depends only on the *time difference*.

$$\begin{aligned} \forall t : \mu_X(t) &= \mu_X \\ \forall t_1, t_2 : R_X(t_1, t_2) &= R_X(t_2 - t_1) = R_X(\tau) \end{aligned}$$

Strict sense stationary  $\implies$  wide sense stationary.

### 1.2 Mean and correlation

Defined as expectation of r.v.  $X(t_k)$  by observing process at time  $t_k$ .

$$\mu_X(t_k) = E[X(t_k)] = \int_{-\infty}^{\infty} x f_{\{X(t_k)\}}(x) dx$$

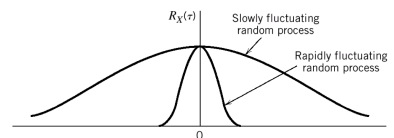
Autocorrelation function  $R_X$  and autovariance function  $C_X$  of a random process:

$$\begin{aligned} R_X(t_1, t_2) &= E[X(t_1)X(t_2)] \triangleq \\ &\int_{-\infty}^{\infty} \int_{-\infty}^{\infty} x_1 x_2 f_{X_1, X_2}(x_1, x_2) dx_1 dx_2 \\ R_{XY}(x, y) &= \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} xy f_{X, Y}(x, y) dx dy \\ C_X(t_1, t_2) &= R_X(t_1, t_2) - m_X^2 \\ &= R_X(t_2 - t_1) - m_X^2 \text{ (for WSS)} \end{aligned}$$

- The mean and autocorrelation function determine the autocovariance function
- The mean and autocorrelation function only describe the first two moments of the process

Properties of the autocorrelation function:

$$\begin{aligned} E[X^2(t)] &= R_X(0) & R_X(\tau) &= R_X(-\tau) \\ |R_X(\tau)| &\leq R_X(0) \end{aligned}$$



The Cross-correlation function  $R_{XY}(t, u)$  of two random processes:

$$\begin{aligned} R_{XY}(t, u) &= E[X(t)Y(u)] = \\ &\int_{-\infty}^{\infty} xy \cdot f_{X, Y}(x, y) dx dy \end{aligned}$$

- Stationary means  $R_{XY}(t, u) = R_{XY}(\tau)$  for  $\tau = t - u$
- Not generally an even function of  $t$
- Not necessarily a maximum at  $\tau = 0$
- Symmetry:  $R_{XY}(\tau) = R_{YX}(-\tau)$

### 1.3 Ergodicity

Definition: A random process is *ergodic* in the mean if

- Time average approaches ensemble averages for increasing  $T$
- The variance of the time average approaches zero for incr.  $T$

$$\lim_{T \rightarrow \infty} \mu_X(T) = \mu_X \quad \lim_{T \rightarrow \infty} \text{Var}[\mu_X(T)] = 0$$

Or in other words: The same behavior averaged over time as averaged over the space of all the system's states.

### 1.4 Filtered processes

Stationary random process  $X(t)$  is input to a linear timeinvariant (LTI) filter with impulse response  $h(t)$ .

$$\begin{aligned} Y(t) &= \int_{-\infty}^{\infty} h(\tau_1) X(t - \tau_1) d\tau_1 \\ S_Y(f) &= |H(f)|^2 S_X(f) \end{aligned}$$

Find mean and autocorrelation of  $Y(t)$ :

$$\begin{aligned} \mu_Y &= E[Y(t)] & R_Y(\tau) &= E[Y(t)Y(t-\tau)] \\ \mu_Y &= E[Y(t)] = E \left[ \int_{-\infty}^{\infty} h(\tau_1) X(t - \tau_1) d\tau_1 \right] \end{aligned}$$

Can interchange expectation and integration if stable  $\int_{-\infty}^{\infty} |h(t)| dt < \infty$  and finite mean  $\mu_X < \infty$

$$\mu_Y = \int_{-\infty}^{\infty} h(\tau_1) E[X(t - \tau_1)] d\tau_1 = \mu_X \int_{-\infty}^{\infty} h(\tau_1) d\tau_1$$

Autocorrelation:

$$\begin{aligned} R_Y(t, u) &= E[Y(t)Y(u)] = \\ E \left[ \int_{-\infty}^{\infty} h(\tau_1) X(t - \tau_1) d\tau_1 \int_{-\infty}^{\infty} h(\tau_2) X(u - \tau_2) d\tau_2 \right] \end{aligned}$$

Additional condition for interchange is finite mean-square value:  $R_X(0) = E[X^2(t)] < \infty$

$$R_Y(\tau) = \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} h(\tau_1) h(\tau_2) R_X(\tau - \tau_1 + \tau_2) d\tau_1 d\tau_2$$

(WS) stationary input process  $X(t)$  to a stable LTI filter  $\implies$  (WS) stationary output process  $Y(t)$ .

### 1.5 Power spectral density

$$S_X(f) = \mathcal{F}[R_X(\tau)](f) = \int_{-\infty}^{\infty} R_X(\tau) e^{-j2\pi f\tau} d\tau$$

- $S_X(0) = \int_{-\infty}^{\infty} R_X(\tau) d\tau$
- $E[X^2(t)] = R_X(0) = \int_{-\infty}^{\infty} S_X(f) df$
- $S_X(f) \geq 0 \forall f$
- $S_X(f) = S_X(-f) \forall f$ , iff  $X(t) \in \mathbb{R}$

### 1.6 Gaussian process

Consider the r.v.  $Y = \int_0^T g(t)X(t)dt$  where  $g(t)$  is in an arbitrary function. If  $Y$  is gaussian distributed, then the process  $X(t)$  is a *Gaussian process*

- A filtered Gaussian process remains a Gaussian process
- If  $X(t)$  is a GP, the arbitrary set of r.v.  $\vec{X} = [X(t_1), \dots, X(t_n)]^T$  is jointly gaussian distributed for any  $n$
- The joint cdf is of these r.v. is completely determined by the **means**  $\mu_X(t_i) = E[X(t_i)]$  and **covariances**  $C_X(t_k, t_i) = E[(X(t_k) - \mu_X(t_k))(X(t_i) - \mu_X(t_i))]$

Multivariate Gauss distribution:

$$\begin{aligned} f(x) &= \frac{\exp(-\frac{1}{2}(\vec{x} - \vec{m}_x)^T \underline{\Sigma}^{-1}(\vec{x} - \vec{m}_x))}{(2\pi)^{\frac{n}{2}} \det(\underline{\Sigma})^{\frac{1}{2}}} \\ \underline{\Sigma} &:= \begin{bmatrix} \text{Cov}(X_1, X_1) & \dots & \text{Cov}(X_1, X_n) \\ \vdots & \ddots & \vdots \\ \text{Cov}(X_n, X_1) & \dots & \text{Cov}(X_n, X_n) \end{bmatrix} \end{aligned}$$



- Channel adds AWGN  $x(t) = s_i(t) + w(t)$  for  $0 \leq t \leq T$
- The optimal receiver minimizes the avg. prob. of symbol error  $P_e$

$$P_e = \sum_{i=1}^M p_i \mathbf{P}(\hat{m} \neq m_i | m_i)$$

### 3.1 Geometric Signal Representation

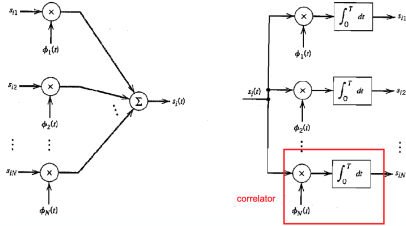
Let  $\{\phi_i(t)\}_{i=1 \dots N}$  be a set of orthonormal basis functions of the signal set  $\{s_i(t)\}_{i=1 \dots M}$ . All signals can be expressed as a finite sum. The coeff.  $s_{ij}$  are given by the projection onto  $\{\phi_i(t)\}_{i=1 \dots N}$ .

The orthonormal functions define a  $N$ -dimensional Euclidean space - the signal space.

$$\int_0^T \phi_i(t) \phi_j(t) dt = \delta_{ij} = \begin{cases} 1, & i = j \\ 0, & -j \end{cases}$$

$$s_i(t) = \sum_{j=1}^N s_{ij} \phi_j(t) \quad s_{ij} = \int_0^T s_i(t) \phi_j(t) dt$$

$$0 \leq t \leq T, \quad i = 1 \dots M, \quad j = 1 \dots N$$



$$\langle s_i(t), s_k(t) \rangle = \int_0^T s_i(t) s_k(t) dt = \mathbf{s}_i^\top \cdot \mathbf{s}_k$$

$$\|\mathbf{s}_i\|^2 = \langle s_i(t), s_i(t) \rangle = \int_0^T s_i(t)^2 dt$$

$$\|\mathbf{s}_i - \mathbf{s}_k\|^2 = \sum_{j=1}^N (s_{ij} - s_{kj})^2 = \int_0^T (s_i(t) - s_k(t))^2 dt$$

$$\cos \theta_{jk} = \frac{\mathbf{s}_i^\top \cdot \mathbf{s}_k}{\|\mathbf{s}_i\| \cdot \|\mathbf{s}_k\|} \quad E_i = \sum_{j=1}^N s_{ij}^2 = \|\mathbf{s}_i\|^2$$

**Gram-Schmidt orthogonalization procedure:** Start with a complete system  $s_1(t), \dots, s_M(t)$  that generates the signal space. At each step generate a new basis function  $\phi_i$ . The basis has only  $N \leq M$  functions.

- Build basis function  $\phi_1$  from  $s_1$

$$\phi_1(t) = \frac{s_1(t)}{\sqrt{\int_0^T s_1^2(t) dt}}$$

- Search for a basis function from  $s_2(t)$

$$s_{21} = \langle s_2(t), \phi_1(t) \rangle = \int_0^T s_2(t) \phi_1(t) dt$$

$$g_2(t) = s_2(t) - s_{21} \phi_1(t)$$

If  $g_2 = 0$ ,  $s_2$  is lin. dep. on  $\phi_1$  and does not lead to a new basis function. Otherwise:

$$\phi_2(t) = \frac{g_2(t)}{\sqrt{\int_0^T g_2^2(t) dt}}$$

- Search for a basis function from  $s_3(t)$

$$s_{31} = \langle s_3(t), \phi_1(t) \rangle \quad s_{32} = \langle s_3(t), \phi_2(t) \rangle$$

$$g_3(t) = s_3(t) - s_{31} \phi_1(t) - s_{32} \phi_2(t)$$

If  $g_3 = 0$ ,  $s_3$  is lin. dep. on  $\phi_1$  and  $\phi_2$  and does not lead to a new basis function. Otherwise:

$$\phi_3(t) = \frac{g_3(t)}{\sqrt{\int_0^T g_3^2(t) dt}}$$

- Search for a basis function from  $s_M(t)$ . Project  $s_M$  on the already determined basis functions, decompose  $s_M$  into its projection and a difference term  $g_M$ . If  $g_M \neq 0$ :

$$\phi_N(t) = \frac{g_M(t)}{\sqrt{\int_0^T g_M^2(t) dt}}$$

### 3.2 Discrete System Model

The signal vector  $\mathbf{s}$ , noise vector  $\mathbf{w}$  and the received signal  $\mathbf{x}$ .

$$\mathbf{s}_i = [s_{i1} \quad \dots \quad s_{iN}]^\top \quad \mathbf{w} = [w_1 \quad \dots \quad w_N]^\top$$

$$\mathbf{x} = [x_1 \quad \dots \quad x_N]^\top = \mathbf{s}_i + \mathbf{w}$$

$$\mathbf{E}[w_j] = 0 \quad \mathbf{E}[w_j \cdot w_k] = \delta_{jk} \quad \text{Var}(w_j) = \frac{N_0}{2}$$

**Theorem of Irrelevance** For signal detection with AWGN, only the projection of the noise onto

the basis functions of the signal set  $\{s_i(t)\}_{i=1}^M$  affect the sufficient statistics of the detection problem. The remainder of the noise is irrelevant.

$$\mu_{X_j} = \mathbf{E}[X_j] = \mathbf{E}[s_{ij} + W_j] = s_{ij} + \mathbf{E}[W_j] = s_{ij}$$

$$\sigma_{X_j}^2 = \text{Var}(X_j) = \mathbf{E}[(X_j - s_{ij})^2] = \mathbf{E}[W_j^2] = \frac{N_0}{2}$$

$$W_j = \int_0^T W(t) \phi_j(t) dt$$

The elements  $X_j$  and  $X_k$  of the received signal vector have the covariance

$$\text{Cov}(x_j, x_k) = \mathbf{E}[(x_j - \mu_{x_j})(x_k - \mu_{x_k})] = 0, \quad j \neq k$$

Thus the  $x_j$  are mutually uncorrelated.  $\Rightarrow$  statistical independence.

**Likelihood Function** As the  $x_j$  are statistically indep. the conditional PDF of  $\mathbf{x}$  given  $\mathbf{s}$  (i.e. symbol  $m_i$  sent using signal  $s_i$ ) follows:

$$L(\mathbf{s}_i) := f_{\mathbf{x}}(\mathbf{x} | \mathbf{s}_i) = f_{\mathbf{W}}(\mathbf{w} = \mathbf{x} - \mathbf{s}_i) =$$

$$= \frac{1}{(\pi N_0)^{N/2}} \exp \left[ -\frac{1}{N_0} \sum_{j=1}^N (x_j - s_{ij})^2 \right]$$

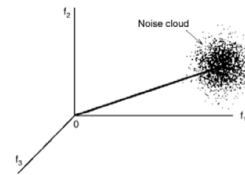
$$l(\mathbf{s}_i) = \log L(\mathbf{s}_i) = -\frac{1}{N_0} \sum_{j=1}^N (x_j - s_{ij})^2 + c$$

$$c = -\frac{N}{2} \log(\pi N_0) \quad i \in \{1, \dots, M\}$$

$L$  likelihood function,  $l$  log-likelihood function can be used because the pdf is always nonnegative and monot. incr. The constant  $c$  is indep. of hyp.  $\mathbf{s}_i$  and can be discarded for the decision.

### 3.3 Detection and Decoding

Detection problem: Given the observation  $\mathbf{x}$ , determine an estimate  $\hat{m}$  of the transmitted symbol  $m_i$ , s.t. the probability of error is minimized.



$$P_e(m_i | \mathbf{x}) = \mathbf{P}(m_i \text{ not sent} | \mathbf{x}) = 1 - \mathbf{P}(m_i \text{ sent} | \mathbf{x})$$

The MAP (Maximum-A-Posteriori) decision rule is optimum in the minimum prob. of error sense. Set  $\hat{m} = m_i$  if:

$$\mathbf{P}(m_i \text{ sent} | \mathbf{x}) \geq \mathbf{P}(m_k \text{ sent} | \mathbf{x}) \quad \forall k \neq i$$

Rephrased using Baye's rule, set  $\hat{m} = m_i$  if  $(p_k \text{ a priori prob. of transmitting } m_k, f_{\mathbf{x}}(\mathbf{x} | m_k) \text{ cond. pdf of } \mathbf{x} \text{ given } m_k)$ :

$$\hat{m} = \arg \max_{m_k} \frac{p_k \cdot f_{\mathbf{x}}(\mathbf{x} | m_k)}{f_{\mathbf{x}}(\mathbf{x})} \quad \forall k \neq i$$

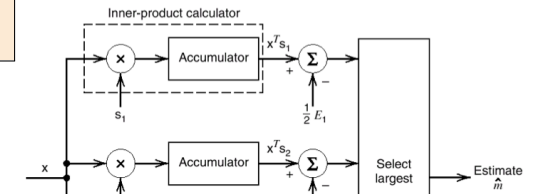
We can drop  $f_{\mathbf{x}}(\mathbf{x})$  as it is indep. of the symbol decision. For equiprobable source symbols, we obtain the ML decision rule: Set  $\hat{m} = m_i$  if  $l(m_k)$  max. for  $k = i$ .

**Simplified ML Rule:**  $\mathbf{x}$  lies in region  $Z_i$  if

$$\sum_{j=1}^N x_j s_{kj} - \frac{1}{2} E_k$$

is maximum for  $k = i$ .

### Correlation receiver



### 3.4 Probability of Error

$\mathbf{P}(A_{ik}) = P_2(\mathbf{s}_i, \mathbf{s}_k)$  is the pairwise error prob. that observation  $\mathbf{x}$  is closer to  $\mathbf{s}_k$  than to  $\mathbf{s}_i$ :

$$P_2(\mathbf{s}_i, \mathbf{s}_k) = \mathbf{P}(\|\mathbf{x} - \mathbf{s}_k\|^2 < \|\mathbf{x} - \mathbf{s}_i\|^2)$$

With the euclidean distance  $d_{14} := \|\mathbf{s}_1 - \mathbf{s}_4\|$ :

$$P_2(\mathbf{s}_1, \mathbf{s}_4) = \mathbf{P}\left(z < -\frac{1}{2} d_{14}\right) = Q\left(\frac{d_{14}}{\sqrt{2N_0}}\right)$$

$$= \frac{1}{2} \text{erfc}\left(\frac{d_{14}}{2\sqrt{N_0}}\right)$$

The pairwise probability of error only depends on the Euclidean distance and is e.g. invariant to rotation and translation of the signal constellation

From the union bound we have

$$P_e(m_i) \leq \sum_{\substack{k=1 \\ k \neq i}}^M P_2(s_i, s_k)$$

$P_e$  is the error prob. averaged over all symbols. An upper bound follows as

$$P_e = \sum_{i=1}^M p_i P_e(m_i) \leq \frac{1}{2} \sum_{i=1}^M \sum_{\substack{k=1 \\ k \neq i}}^M p_i \operatorname{erfc} \left( \frac{d_{ik}}{2\sqrt{N_0}} \right)$$

## 4 Passband Data Transmission

In bandpass data transmission, information modulates a carrier and occupies a restricted bandwidth in frequency. The carrier can be modulated by changing:

- Amplitude (ASK)
- Phase (PSK)
- Frequency (FSK)

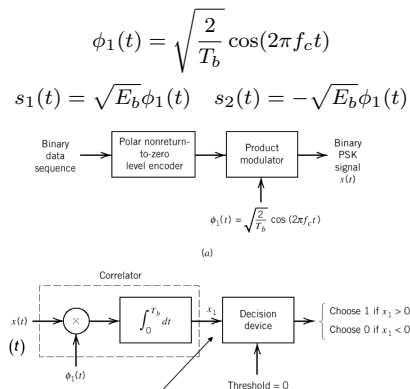
**Coherent** modulation is when the receiver's local oscillator is phase-synchronous to the transmitter's local oscillator.

$M = 2^n$  levels for signalling information ( $M$ -ary xSK). Using  $M$  levels, symbol duration  $T = nT_b$  is changed while keeping the same datarate. Bandwidth shrinks accordingly by  $1/nT_b$ .

Figures of merit: Symbol error probability at given SNR, power spectral density, bandwidth efficiency  $\rho = R_b/B$  [bit/s/Hz].

### 4.1 PSK: Coherent Phase Shift Keying

#### BPSK: Binary PSK



$$p_{10} = p_{01} = P_e = \frac{1}{2} \operatorname{erfc} \left( \sqrt{\frac{E_b}{N_0}} \right)$$

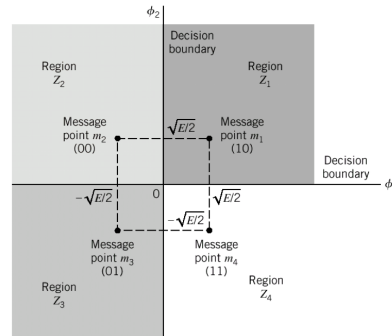
$$S(f - f_c) \approx 2E_b \operatorname{sinc}^2(T_b f)$$

**QPSK: Quadruphase SK**, use more than just two phase levels.

$$\phi_1(t) = \sqrt{\frac{2}{T_b}} \cos(2\pi f_c t) \quad \phi_2(t) = \sqrt{\frac{2}{T_b}} \sin(2\pi f_c t)$$

$$\mathbf{s}_{10} = \begin{bmatrix} +c \\ +c \end{bmatrix} \quad \mathbf{s}_{00} = \begin{bmatrix} -c \\ +c \end{bmatrix} \quad \mathbf{s}_{01} = \begin{bmatrix} -c \\ -c \end{bmatrix} \quad \mathbf{s}_{11} = \begin{bmatrix} +c \\ -c \end{bmatrix}$$

$$c = \sqrt{E/2}$$



Every QPSK symbol carries 2 bits, hence the symbol energy is twice the energy per information bit:  $E = 2E_b$ . A QPSK system achieves same BER ( $P_e$ ) as a BPSK at same  $E_b/N_0$  but at *twice the bit rate*.

$$P_e = \frac{1}{2} \operatorname{erfc} \left( \sqrt{\frac{E_b}{N_0}} \right)$$

$$S_B(f) = 4E_b \operatorname{sinc}^2(2T_b f)$$

### 4.2 QAM: Hybrid Amplitude/Phase Modulation

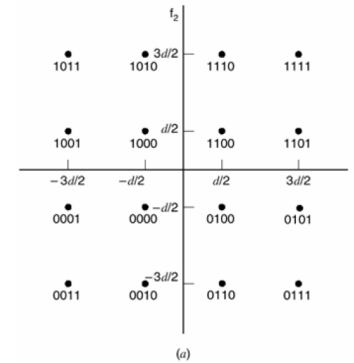
**QAM: M-ary quadrature amplitude modulation**, change phase and amplitude.

$d_{\min}$  is the distance between adjacent messages in the signal space.

$$\phi_1(t) = \sqrt{\frac{2}{T_b}} \cos(2\pi f_c t) \quad \phi_2(t) = \sqrt{\frac{2}{T_b}} \sin(2\pi f_c t)$$

$$\mathbf{s}_i = \frac{d_{\min}}{2} \begin{bmatrix} a_i \\ b_i \end{bmatrix} \quad a_i, b_i \text{ odd integers, } i = 1, \dots, M$$

Mapping an even number  $f$  bits per symbol (e.g. 4bits  $\rightarrow$  16 symbols), results in a quadratic  $L \times L$  square constellation with  $L = \sqrt{M}$ . Gray coding is often used for mapping the bits to the QAM symbols.



$$P_e = (1 - P'_e)^2 \rightarrow P_e = 1 - P_c = 1 - (1 - P'_e)^2 \approx 2P'_e$$

$$P'_e = \left( 1 - \frac{1}{\sqrt{M}} \right) \operatorname{erfc} \left( \sqrt{\frac{d_{\min}^2}{4N_0}} \right)$$

$$E_{av} = \frac{(M-1)d_{\min}^2}{6}$$

$$P_e \approx 2P'_e = 2 \left( 1 - \frac{1}{\sqrt{M}} \right) \operatorname{erfc} \left( \sqrt{\frac{3E_{av}}{2(M-1)N_0}} \right)$$

$E_{av}$  average symbol energy.

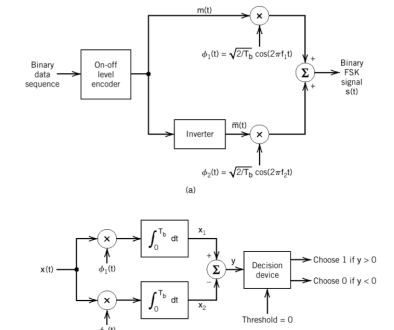
### 4.3 FSK: Coherent Frequency-Shift Keying

$$\phi_i(t) = \begin{cases} \sqrt{\frac{2}{T_b}} \cos(2\pi f_i t), & 0 \leq t \leq T_b \\ 0, & \text{else.} \end{cases}$$

$$f_i = \frac{n_c + i}{T_b} \quad i = 1, 2, \quad n_c \in \mathbb{N}$$

$$\mathbf{s}_1 = \sqrt{E_b} \begin{bmatrix} 1 \\ 0 \end{bmatrix} \quad \mathbf{s}_2 = \sqrt{E_b} \begin{bmatrix} 0 \\ 1 \end{bmatrix}$$

$f_i$  chosen by rule to avoid phase discontinuities. The two frequencies  $f_1$  and  $f_2$  are  $1/T_b$  Hz apart. The  $\phi_i$  are orthogonal for  $f_i = (n_c + i)/T_b$ .



Distance between message points in signal space is  $1/\sqrt{2}$  smaller compared to binary PSK.

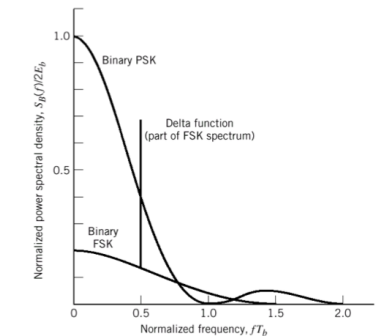
$$d_{\min} = \sqrt{2E_b}$$

$$P_e = \frac{1}{2} \operatorname{erfc} \left( \sqrt{\frac{(d_{\min}/2)^2}{N_0}} \right) = \frac{1}{2} \operatorname{erfc} \left( \sqrt{\frac{E_b}{2N_0}} \right)$$

$$S_B(f) = \frac{E_b}{2T_b} \left[ \delta \left( f - \frac{1}{2T_b} \right) + \delta \left( f + \frac{1}{2T_b} \right) \right] + \dots$$

$$\frac{8E_b \cos^2(\pi T_b f)}{\pi^2 (4T_b^2 f^2 - 1)^2}$$

PSD contains two delta pulses and decays much faster than BPSK due to continuous phase operation.



### 4.4 CPFSK Continuous Phase FSK

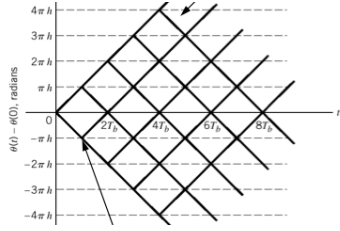
$$s(t) = \sqrt{\frac{2E_b}{T_b}} \cos(2\pi f_c t + \theta(t))$$

$$\theta(t) = \theta(0) \pm \frac{\pi h}{T_b} t, \quad 0 \leq t \leq T_b$$

$$h = T_b(f_1 - f_2), \quad f_c = \frac{1}{2}(f_1 + f_2)$$



$h$  modulation index.



#### 4.5 MSK Minimum Shift Keying

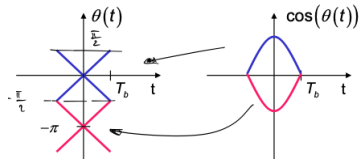
For integer valued  $h$ , the accumulated phase at end of symbol is independent of the previous and current symbol.  $\Rightarrow$  no phase memory, each symbol can be decoded independently.

$$f_1 - f_2 = 0.5/T_b$$

The minimum difference, for which  $s_1(t), s_2(t)$  orthogonal.

$\theta(0) = 0$	$\theta(T_b) = \pi/2$	symbol 1 transmitted
$\theta(0) = \pi$	$\theta(T_b) = \pi/2$	symbol 0 transmitted
$\theta(0) = -\pi$	$\theta(T_b) = -\pi/2$	symbol 1 transmitted
$\theta(0) = 0$	$\theta(T_b) = -\pi/2$	symbol 0 transmitted

**Estimation of  $\theta(0)$ :** Expanding  $s(t)$  into two terms we get:



We can estimate  $\theta(0)$  by observing

$$\sqrt{\frac{2E_b}{T_b}} \cos(\theta(t)) \cos(2\pi f_c t)$$

#### MSK Signal-Space representation

$$\phi_1(t) = \sqrt{\frac{2}{T_b}} \cos\left(\frac{\pi}{2T_b}t\right) \cos(2\pi f_c t), \quad -T_b \leq t \leq T_b$$

$$\phi_2(t) = \sqrt{\frac{2}{T_b}} \sin\left(\frac{\pi}{2T_b}t\right) \sin(2\pi f_c t), \quad 0 \leq t \leq 2T_b$$

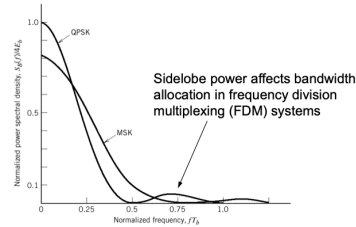
A coherent receiver has to integrate over two bit periods:

$$x_1 = \int_{-T_b}^{T_b} x(t) \phi_1(t) dt = \sqrt{E_b} \cos(\theta(0)) + w_1$$

$$x_2 = \int_0^{2T_b} x(t) \phi_2(t) dt = -\sqrt{E_b} \sin(\theta(T_b)) + w_2$$

**Bit error rate** The four points in the signal-space diagram correspond to two symbol, hence the BER ( $P_e$ ) is the same as with QPSK.

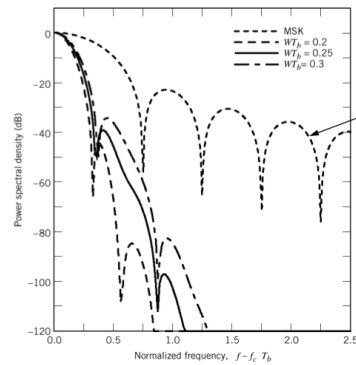
$$P_e = \frac{1}{2} \operatorname{erfc}\left(\frac{d_{\min}/2}{\sqrt{N_0}}\right) = \frac{1}{2} \operatorname{erfc}\left(\sqrt{\frac{E_b}{N_0}}\right)$$



#### 4.6 GMSK

To make sidelobes of MSK smaller, filter the NRZ signal with pulse shaping function.

$$H(f) = \exp\left[-\frac{\log 2}{2} \left(\frac{f}{W}\right)^2\right]$$

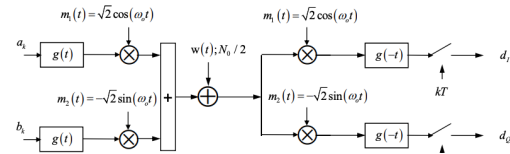


The parameter  $\alpha$  depends on the time bandwidth product  $WT_b$ . The quantity  $10 \log(\alpha/2)$  expresses the degradation in dB of GMSK compared to MSK. MSK:  $WT_b = \infty, \alpha = 2$

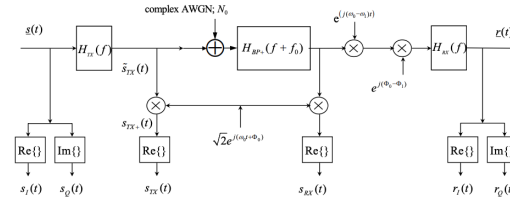
$$P_e = \frac{1}{2} \operatorname{erfc}\left(\sqrt{\frac{\alpha E_b}{2N_0}}\right)$$

#### 4.7 Equivalent baseband representation

**QAM:** Two branches: inphase (I) and quadrature (Q)



By using complex valued signals, the transmission system can be written as an LTI system.



Important names and notation:

$\tilde{s}_{TX}(t)$	complex envelope of $s_{TX}$
$s_{TX+}(t)$	analytic signal (pre-envelope of $s_{TX}$ )
$s_{TX}(t)$	physical passband signal

$$s_{TX+}(f) = \begin{cases} 2s_{TX}(f) & f > 0 \\ 0 & \text{else} \end{cases}$$

$$H_{BP+}(f) = \begin{cases} H_{BP}(f) & f > 0 \\ 0 & \text{else} \end{cases}$$

$$H_{TX}(f) = 0 \forall |f| \geq f_0 \quad H_{RX}(f) = 0 \forall |f| \geq B$$

$$B = \max(0, f_0 - |f_0 - f_1|)$$

**Complex Envelope** Inphase and quadrature components.

$$\tilde{s}_{TX}(t) = \tilde{s}_{TX,I}(t) + j\tilde{s}_{TX,Q}(t) \quad \tilde{S}_{TX}(f) = 0 \forall |f| > f_0$$

**Analytic signal** Pre-envelope. Has one-sided spectrum, scaling factor to preserve power values in passband and equivalent baseband.

$$s_{TX+}(t) = \tilde{s}_{TX}(t) \sqrt{2} \exp(j\omega_0 t) \exp(j\phi_0)$$

$$S_{TX+}(f) = \sqrt{2} \exp(j\phi_0) \tilde{S}_{TX}(f - f_0)$$

$$S_{TX+}(f) = 0 \forall f < 0$$

$$\tilde{S}_{TX}(f) = \frac{1}{\sqrt{2}} \exp(-j\phi_0) S_{TX+}(f + f_0)$$

#### Physical passband signal

$$s_{TX}(t) = \operatorname{Re}\{s_{TX+}(t)\}$$

$$S_{TX}(f) = \frac{1}{2} (S_{TX+}(f) + S_{TX+}^*(-f))$$

$$s_{TX}(t) = \sqrt{2} \tilde{s}_{TX,I}(t) \cos(\omega_0 t + \phi_0) - \sqrt{2} \tilde{s}_{TX,Q}(t) \sin(\omega_0 t + \phi_0)$$

$$s_{TX}(t) = \left\{ \sqrt{2} \sqrt{\tilde{s}_{TX,I}^2(t) + \tilde{s}_{TX,Q}^2(t)} \right\} \cos(\omega_0 t + \phi_0 + \phi(t))$$

$$\phi(t) = \operatorname{atan2}(\tilde{s}_{TX,Q}(t), \tilde{s}_{TX,I}(t))$$

#### Summary

$$x(t) = \operatorname{Re}\{x_+(t)\} \quad x_+(t) = \tilde{x}(t) \sqrt{2} e^{j2\pi f t}$$

$x(t)$	physical passband signal
$x_+(t)$	analytic signal (pre-envelope of $x(t)$ )
$\tilde{x}(t)$	complex envelope of $x(t)$

#### 4.8 Noncoherent Detection

Carrier phase  $\theta$  at the receiver becomes a random variable.

#### 4.9 ML detection with unknown phase shift

$$L(\mathbf{s}_i) \triangleq f_X(\mathbf{x}|\mathbf{s}_i) =$$

$$\frac{1}{(\pi N_0)^{N/2}} \exp\left[-\frac{1}{N_0} \sum_{j=0}^N (x_j - s_{ij})^2\right]$$

The ML receiver selects the hypothesis, which maximizes the likelihood function

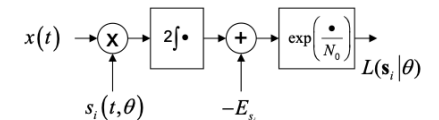
$$\hat{i} = \arg \max_i (L(\mathbf{s}_i))$$

Expanding the sum in the exponent, the likelihood function can be calculated from the output of a correlator bank.

$$L(\mathbf{s}_i) = c \exp\left[\frac{2}{N_0} \int x(t) s_i(t) dt - \frac{1}{N_0} E_{s_i}\right]$$

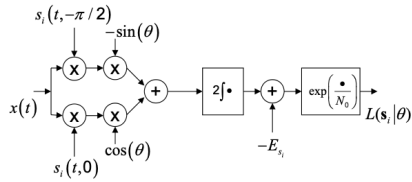
With  $E_{s_i} = \sum_j s_{ij}^2 = \int s_i^2 dt$

For a known phase offset, the modified receiver correlates with a rotate version of each hypothesis.



#### Two-branch correlator

$$s_i(t, \theta) = s_i(t, \theta = 0) \cos \theta - s_i(t, \theta = -\pi/2) \sin \theta$$



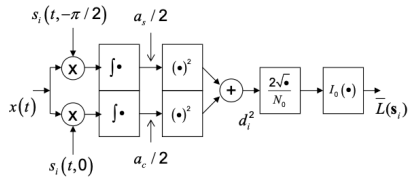
**Equi-Energy Signals with unknown phase offset** Shifting the integrator to each branch and obtain equi-energy signals with known phase offset:

$$L(s_i|\theta) = \exp\left(\frac{1}{N_0}(a_c \cos \theta - a_s \sin \theta)\right) \\ = \exp\left(\frac{1}{N_0}\sqrt{a_c^2 + a_s^2} \cos(\theta + \phi)\right) \\ \phi = \angle(a_c + ja_s)$$

With unknown phase offset, we have to average the likelihood function across all phase offsets  $\theta$ .

$$\overline{L(s_i)} = \frac{1}{2\pi} \int_{-\pi}^{\pi} \exp\left(\frac{1}{N_0}\sqrt{a_c^2 + a_s^2} \cos(\theta + \phi)\right) d\theta \\ = I_0\left(\frac{1}{N_0}\sqrt{a_c^2 + a_s^2}\right)$$

$I_0$  is the modified Bessel function of order zero.



As  $I_0$  is monotonously increasing, a simplified decision rule follows as

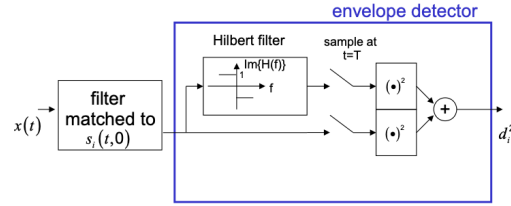
$$\hat{i} = \arg \max_i d_i^2$$

Note that we need a two-branch correlator for each hypothesis  $s_i(t)$ .

Instead of the two-branch correlator we can use two matched filter - sampler pairs to calculate the decision variable.

We can determine the decision variable with one matched filter and a Hilbert transformer. The matched filter - envelope detector pair is called a noncoherent matched filter.

$$s_i(t, \theta = -\pi/2) \circ \bullet S_i(f, \theta = -\pi/2) \\ = -j \operatorname{sgn}(f) S_i(f, \theta = 0)$$

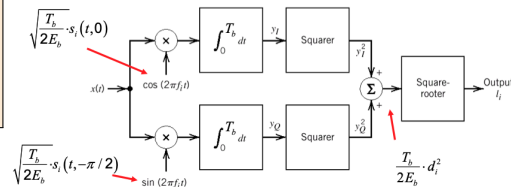


#### 4.10 Noncoherent FSK

Signal  $x(t)$  at the receiver with unknown carrier phase offset  $\theta$ :

$$x(t) = \sqrt{\frac{2E_b}{T_b}} \cos(2\pi f_i t + \theta) + w(t), \quad i = 1, 2, 0 \leq t \leq T_b$$

The signals  $s_1$  and  $s_2$  each require such a branch. A comparator subsequently compares the two outputs  $I_i$  to decide between the hypothesis  $s_1$  and  $s_2$ .

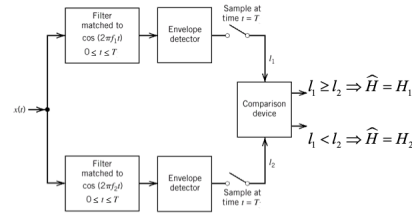


We have

$$P_e = \frac{1}{2} \exp\left(-\frac{E_b}{2N_0}\right)$$

This corresponds to a degradation of at least 3dB compared to coherent MSK. Less degradation compared to BFSK.

Another implementation is with matched bandpass filters to  $f_1$  and  $f_2$  followed by envelope detectors, samplers and a comparison device.

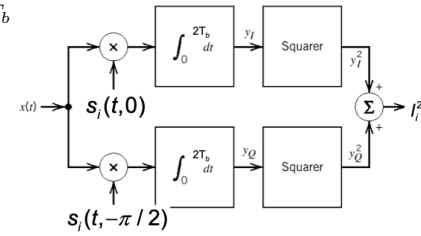


#### 4.11 DPSK Differential PSK

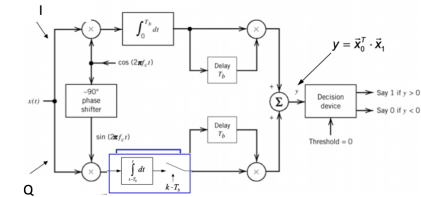
Differential precoding at transmitter: Symbol 0  $\Rightarrow \pi$  phasejump, Symbol 1  $\Rightarrow$  no phase-jump. Assumption:  $\theta$  does not change significantly between two adjacent sampling instances.

$$s_1(t, \theta) = \begin{cases} \sqrt{\frac{2E_b}{T_b}} \cos(2\pi f_c t + \theta), & 0 \leq t \leq T_b \\ \sqrt{\frac{2E_b}{T_b}} \cos(2\pi f_c t + \theta), & T_b \leq t \leq 2T_b \end{cases} \\ s_2(t, \theta) = \begin{cases} \sqrt{\frac{2E_b}{T_b}} \cos(2\pi f_c t + \theta), & 0 \leq t \leq T_b \\ \sqrt{\frac{2E_b}{T_b}} \cos(2\pi f_c t + \pi + \theta), & T_b \leq t \leq 2T_b \end{cases}$$

Noncoherent detector for DPSK:



Quadrature implementation of simplified detector:

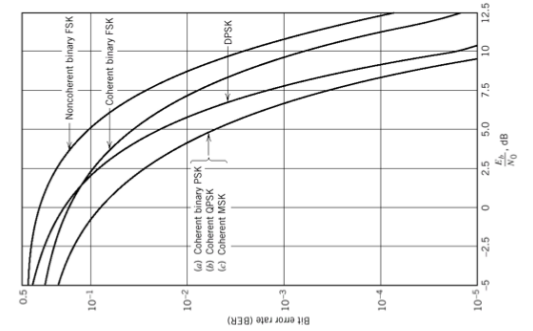


DPSK is a special case of noncoherent, orthogonal modulation with  $T = 2T_b$  and  $E = 2E_b$ . The bit error rate is given by:

$$P_e = \frac{1}{2} \exp\left(-\frac{E_b}{N_0}\right)$$

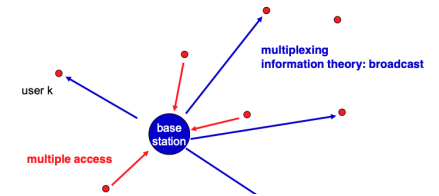
#### 4.12 Performance comparison

Modulation	$P_e$
Coherent BPSK	$\frac{1}{2} \operatorname{erfc}\left(\sqrt{\frac{E_b}{N_0}}\right)$
Coherent QPSK	
Coherent MSK	
Coherent binary FSK	$\frac{1}{2} \operatorname{erfc}\left(\sqrt{\frac{E_b}{2N_0}}\right)$
DPSK	$\frac{1}{2} \exp\left(-\frac{E_b}{N_0}\right)$
Noncoherent binary FSK	$\frac{1}{2} \exp\left(-\frac{E_b}{2N_0}\right)$



## 5 Multi User Radio Communications

### 5.1 Multiple Access techniques



Accommodation of several users in the same wireless environment.

- FDMA Frequency domain multiple access
- TDMA Time domain multiple access
- CDMA Code division multiple access
- SDMA Spatial division multiple access

### 5.2 Radio Communication over line-of-sight (LOS)

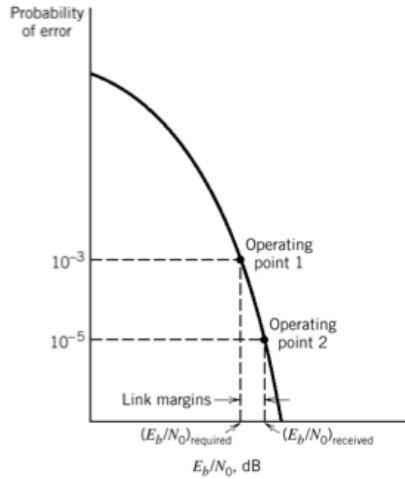
Free-space (line of sight) communication on up- and down-link. AWGN model appropriate. Used MA techniques:

- **FDMA** Non linearity of transponder causes interference between frequency subband. Transponder is operated in lin. regime below max output power. Reduced power efficiency.
- **TDMA** Can operate at close to full power efficiency. Commonly used.
- **SDMA** Multiple antennas allow beam forming to different locations.

**Link budget** Link (power) budget: Budgeting of all gains and losses. Accounting of resources available to transmitter and receiver, sources of loss of power, sources of noise. Allows performance estimation of LOS links.

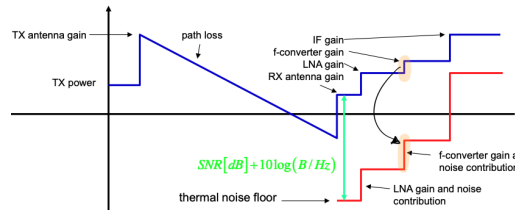
**Link Margin** Curve relates  $P_e$  to  $E_b/N_0$ . Max accepted  $P_e$  leads to Op1 and minimal required

$(E_b/N_0)_{\text{req}}$ . Actually received  $(E_b/N_0)_{\text{rec}}$  define Op2.



Link margin: Difference. Provides Protection against unexpected changes. Large margin: high reliability but low efficiency.

$$M = 10 \log \left( \frac{E_b}{N_0} \right)_{\text{rec}} - 10 \log \left( \frac{E_b}{N_0} \right)_{\text{req}} \quad [M] = \text{dB}$$



### 5.3 Antenna characterization

Receiver is located in the farfield of the transmitter.  $D$  largest dimension of antenna.

$$d_f \gg \frac{2D^2}{\lambda}$$

Idealized reference antenna radiates uniformly in all directions. Power density as a function of distance scales according to free-space propagation.  $\Phi$  Radiation intensity in watts per unit solid angle.

$$\rho(d) = \frac{P_t}{4\pi d^2} \quad [\rho] = \frac{\text{W}}{\text{m}^2} \quad \Phi = d^2 \rho(d)$$

Total radiated power  $P$  and average power per unit

solid angle  $P_{av}$ :

$$P = \int \Phi(\theta, \phi) d\Omega \quad P_{av} = \frac{1}{4\pi} \int \Phi(\theta, \phi) d\Omega = \frac{P}{4\pi}$$

$$[P] = \text{W} \quad [P_{av}] = \frac{\text{W}}{\text{steradian}}$$

**Directivity gain**  $g(\theta, \phi)$  Ratio of radiation intensity in a specific direction to the avg. radiated power.

**Directivity**  $D$ , maximal directivity gain over all directions

**Power gain**  $G$  with  $\eta_{\text{radiation}} \in [0, 1]$  the radiation efficiency factor.

**EIRP** Effective isotropically radiated power referenced to an isotropic source.

**Beamwidth** Angle between the two directions in which the radiation intensity is one-half the maximum. Higher power gain leads to narrower beamwidth.

**Effective Apperture**  $A$  Ratio of power available at the antenna terminals to the power per unit area of the appropriately polarized incident electromagnetic wave.

**Apperture efficiency**  $\eta_{ap}$  with  $A_{ph}$  physical area.

$$g(\theta, \phi) = \frac{\Phi(\theta, \phi)}{P_{av}} = \frac{\Phi(\theta, \phi)}{P/(4\pi)} \quad D = \max_{\theta, \phi} g(\theta, \phi)$$

$$G = \eta_{\text{radiation}} D \quad \text{EIRP} = P_t G_t$$

$$A = (\lambda^2/4\pi) G \quad \eta_{ap} = A/A_{ph}$$

**Frii's Free-Space Equation** Power captured by receiver at distance  $d$  in LoS:

$$P_r = \left( \frac{\text{EIRP}}{4\pi d^2} \right) A_r = \frac{P_t G_t A_r}{4\pi d^2} = P_t G_t G_r \left( \frac{\lambda}{4\pi d} \right)^2$$

**Path loss** PL is the difference between transmit signal power and receive signal power.

$$\text{PL} = 10 \log \frac{P_t}{P_r} = -10 \log(G_t G_r) + 10 \log \left( \frac{4\pi d}{\lambda} \right)^2$$

### 5.4 Noise figure

Spot noise figure  $F(f)$  ratio of total available output noise power per unit bandwidth to portion thereof due to source alone.  $G$  power gain. **Lower is better**

$$F(f) = \frac{S_{NO}(f)}{G(f)S_{NS}(f)} = \frac{P_S S_{NO}}{P_O S_{NS}} = \frac{\text{SNR}_{\text{Source}}(f)}{\text{SNR}_{\text{Output}}(f)}$$

Signal to noise ratio after receiver amplifier is

$$\text{SNR}_{\text{in}} - F$$

If two-port is noise free:

$$S_{NO}(f) = G(f)S_{NS}(f)$$

For physical systems

$$S_{NO}(f) > G(f)S_{NS}(f)$$

**Equivalent Noise Temperature**  $T_e$ . For low noise devices  $T_e$  is a better measure bcs  $F$  is close to unity. Two-port device matched to source impedance is considered.  $N_1$  available input noise power:

$$N_1 = \left( \frac{\sqrt{4kTR_sB}}{2R_s} \right)^2 R_s = kTB.$$

Total output noise power  $N_2$  and noise figure:

$$N_2 = GN_1 + N_d = Gk(T + T_e)B$$

$$F = \frac{N_2}{N_2 - N_d} = \frac{T + T_e}{T}$$

Equivalent noise temperature  $T_e$

$$T_e = T(F - 1)$$

**Cascade of Two-Port Networks** Use factor not dB! Best if Lowest  $F$  first in chain.

$$F = F_1 + \frac{F_2 - 1}{G_1} + \frac{F_3 - 1}{G_1 G_2} + \frac{F_4 - 1}{G_1 G_2 G_3} + \dots$$

$$T_e = T_{e1} + \frac{T_{e2}}{G_1} + \frac{T_{e3}}{G_1 G_2} + \frac{T_{e4}}{G_1 G_2 G_3} + \dots$$

### 5.5 Radio Communication over multipath channels

In mobile radio systems the transmitters and receivers are mobile. This leads to a stochastic channel. Multipath phenomenon leads to **Fading**.

**Narrowband fading** Complex envelopes of  $\tilde{s}$  Tx,  $\tilde{s}_o$  Rx signal and  $\tilde{h}$  timevarying impulse response of channel. The Rayleigh fading model for NLOS conditions is modeled as zero-mean complex Gaussian random process. Characterized by autocorrelation function  $R_{\tilde{h}}$  and Doppler spectrum  $S_{\tilde{H}}$ .

$$\tilde{s}_o(t) = \int_{-\infty}^{\infty} \tilde{s}(t-\tau) \tilde{h}(\tau; t) d\tau$$

$$\tilde{h}(\tau; t) = \tilde{h}(t) \delta(\tau) \implies \tilde{s}_o(t) = \tilde{h}(t) \tilde{s}(t)$$

$$R_{\tilde{h}}(\Delta t) = E[\tilde{h}^*(t) \tilde{h}(t + \Delta t)] \circ \bullet S_{\tilde{H}}(\nu)$$

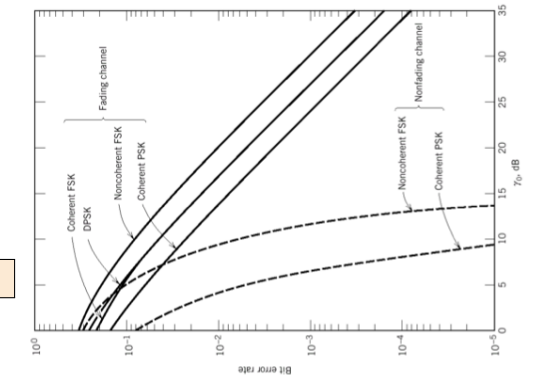
**Coherent BPSK over slow rayleigh fading channel** Input-output relation is  $\tilde{s}_o(t) = \alpha \exp(-j\phi) \tilde{s}(t) + \tilde{w}(t)$  with  $\alpha$  and  $\phi$  Rayleigh and uniformly distributed r.v.

$$P_{e|\tilde{h}}(\gamma) = \frac{1}{2} \text{erfc} \sqrt{\gamma} \quad \gamma = \frac{\alpha^2 E_b}{N_0}$$

Averaging over all channel realizations:

$$P_e(\gamma_0) = \int_0^{\infty} P_{e|\tilde{h}}(\gamma) f(\gamma) d\gamma = \frac{1}{2} \left( 1 - \sqrt{\frac{\gamma_0}{1 + \gamma_0}} \right)$$

$$\gamma_0 = E[\gamma] = \frac{E_b}{N_0} E[\alpha^2]$$



At high SNR deep fades dominate performance  $P_e \propto 1/\text{SNR}$ .

**Diversity** Availability of independently faded copies of the transmit signal at the receiver. Diversity order  $L$  number of available indep. faded



versions of the same signal. At high SNR,  $L$ -th order diversity allows  $P_e \propto 1/SNR^L$

## 5.6 Summary

$$N_0 = kTF \quad P_n = N_0 B \quad \text{SNR}_{\text{dB}} = P_{r,\text{dB}} - P_{n,\text{dB}}$$

$$P_r = P_t G_t G_r \left( \frac{\lambda}{4\pi d} \right)^2$$

$$P_r = RE_{b,\min} M \quad A = G_r \frac{\lambda^2}{4\pi} = \eta_{ap} A_{ph}$$

$N_0$  Noise power spectral density [W/Hz]

$F$  Noise Figure

$B$  Bandwidth [Hz]

$P_n$  Noise power [W]

$P_r/P_t$  Receiver/Transmitter power [W]

$G_r/G_t$  Receiver/Transmitter antenna gain

$d$  Distance between receiver/transmitter

$R$  Datarate

$E_{b,\min}$  Minimum energy per bit

$M$  Link margin

$A$  Effective aperture

$\eta_{ap}$  Aperture efficiency

$A_{ph}$  Physical area of antenna

## 6 Information Theory

Recap:  $p$  Bit error prob. for BNRZ channel and amplitude  $A$ .

$$p = \frac{1}{2} \operatorname{erfc} \left( \sqrt{\frac{A^2}{N_0}} \right)$$

### 6.1 Uncertainty, Information and Entropy

A source emits a message  $S$ .  $S$  is a r.v. taking values in a finite alphabet  $\mathcal{S} = \{s_0, \dots, s_{K-1}\}$ .

$$\mathbf{P}(S = s_k) = p_k \quad \sum_{k=0}^{K-1} p_k = 1$$

The definition of information  $I$  is:

$$I(s_k) \triangleq -\log p_k$$

$$I(s_k) = 0 \quad \text{for } p_k = 1$$

$$I(s_k) \geq 0 \quad \text{for } 0 \leq p_k \leq 1$$

$$I(s_k) > I(s_i) \quad \text{for } p_k < p_i$$

$$I(s_k s_i) = I(s_k) + I(s_i) \quad \text{if stat. indep. } p_{ki} = p_k p_i$$

When  $I$  is specified in bits, logarithms are to base 2.

The definition of entropy  $H$  is:

$$H(S) \triangleq E[I(S)] = \sum_{k=0}^{K-1} p_k I(s_k) = - \sum_{k=0}^{K-1} p_k \log p_k$$

$$0 \leq H(S) \leq \log_2 K$$

$$H(S) = 0 \quad \text{iff } p_k = 1 \text{ for one } k$$

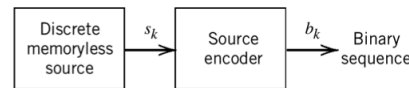
$$H(S) = \log_2 K \quad \text{iff } p_k = \frac{1}{K} \forall k$$

**Extended source** Divide a seq. of  $n \cdot M$  successive source symbols into  $M$  blocks. Consider each block of  $n$  symbols as a single "super symbol" taking on values in  $\mathcal{S}^n$ . The entropy of the extended source is:

$$H(\mathcal{S}^n) = n \cdot H(S)$$

### 6.2 Source Coding Theorem

How is the information of a source efficiently represented? Requirements: Code words must be binary, unique decodability.



The average code word length of a code whose  $k$ th code-word is of length  $L_k$ :

$$\bar{L} = \sum_{k=0}^{K-1} p_k L_k$$

The average code word length is lower bounded by the entropy of the source.

$$\bar{L} \geq H(S) \triangleq L_{\min}$$

Coding efficiency is a measure of code quality:

$$\eta = \frac{L_{\min}}{\bar{L}} \leq 1$$

### 6.3 Data Compaction

Goal: Eliminate redundancy. Seek for codes that approach Shannon's lower bound on the avg code-word length.

**Prefix Codes:** No code-word is a prefix of another code-word. Leads to implicit recognition of end of

code word. For each discrete, memoryless source there exists a prefix code s.t.:

$$H(S) \leq \bar{L} < H(S) + 1$$

$$H(\mathcal{S}^n) \leq \bar{L}_n < H(\mathcal{S}^n) + 1$$

$$\Leftrightarrow nH(S) \leq \bar{L}_n < nH(S) + 1$$

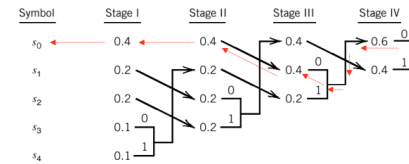
$$\Leftrightarrow H(S) \leq \frac{\bar{L}_n}{n} < H(S) + \frac{1}{n}$$

$\frac{\bar{L}_n}{n}$  effective number of bit per source symbol.

**Huffman Coding** yields a prefix code that minimizes the avg. code-word length when the source is memoryless.

1. Assign a 0 and 1 to the symbols of lowest probability
2. Replace the two symbols by a new pseudo-symbol whose prob. is the sum of the two probs.
3. Repeat 1. and 2. until only a single pseudo-symbol left

The code sequence for each symbol is found by backtracking from last symbol and tracing the 0s and 1s.



Drawbacks:

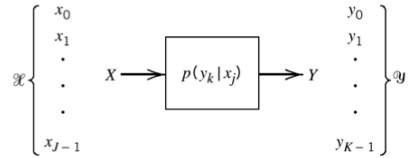
- Need to know probabilities a priori
- Redundancy due to memory in source can only be removed by using large extension codes, increasing complexity

**Lempel-Ziv Coding** Adaptive algorithm of low complexity that captures the source statistic and memory in the source intrinsically.

Numerical Positions:	1	2	3	4	5	6	7	8	9
Subsequences:	0	1	00	01	011	10	010	100	101
Numerical representations:			1-1	1-2	4-2	2-1	4-1	6-1	6-2
Binary encoded blocks:			0010	0011	1001	0100	1000	1100	1101

- Constructed by parsing the source data stream into segments other than 0 and 1 that are shortest subsequences not encountered previously
- Segment 0 and 1 are assigned indices 1 and 2
- $N$  stored subsequences are indexed from 3 to  $N+2$
- A new sequence can always be composed from an old subsequence (root subsequence) and a 0 and a 1 (innovation symbol)

### 6.4 Discrete Memoryless Channel



$X, Y$  are statistically dependant r.v. Discrete if the input and output alphabet  $(\mathcal{X}, \mathcal{Y})$  are of finite size. Memoryless if the current output depends on the current input only. It is fully described by

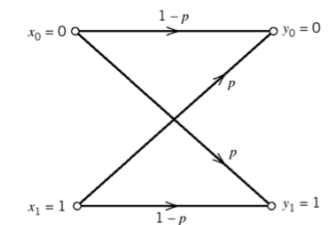
- input alphabet  $\mathcal{X} = \{x_0, \dots, x_{J-1}\}$
- output alphabet  $\mathcal{Y} = \{y_0, \dots, y_{K-1}\}$
- transition probabilities  $p(y_k | x_j) = \mathbf{P}(Y = y_k | X = x_j)$

**Transition Matrix**

$$\mathbf{P} = \begin{bmatrix} p(y_0|x_0) & \dots & p(y_{K-1}|x_0) \\ \vdots & \ddots & \vdots \\ p(y_0|x_{J-1}) & \dots & p(y_{K-1}|x_{J-1}) \end{bmatrix}$$

$$p(y_k) = \mathbf{P}(Y = y_k) = \sum_{j=0}^{J-1} p(y_k | x_j) p(x_j)$$

### Binary, Symmetric Channel



$J = K = 2$ , transition probability  $p$ . Error probability is  $p$ .

$$C_{\text{BSC}} = 1 + p \log_2 p + (1-p) \log_2 (1-p)$$

## 6.5 Mutual Information

**Conditional entropy** of  $X$  given  $Y$  is a measure for the uncertainty about  $X$  if  $Y$  is known:

$$H(X|Y) = -E_{X,Y}[\log_2 p(X|Y)] = -\sum_{k=0}^{K-1} \sum_{j=0}^{J-1} p(x_j, y_k) \log_2 p(x_j|y_k)$$

Remember:  $p(x_j, y_k) = p(x_j|y_k) \cdot p(y_k)$

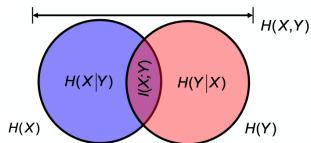
**Mutual information** is the reduction of the uncertainty about  $X$  achieved by observing  $Y$ .

$$I(X; Y) \triangleq H(X) - H(X|Y)$$

$$\begin{aligned} I(X; Y) &= I(Y; X) \\ H(X) - H(X|Y) &= H(Y) - H(Y|X) \\ I(X; Y) &\geq 0 \\ H(X; Y) &\geq H(X|Y) \\ I(X; Y) &= 0 \Leftrightarrow X, Y \text{ independent} \end{aligned}$$

**Joint entropy** of  $X$  and  $Y$  is defined as:

$$\begin{aligned} H(X, Y) &= -E_{X,Y}[\log_2 p(X, Y)] = -\sum_{k=0}^{K-1} \sum_{j=0}^{J-1} p(x_j, y_k) \log_2 p(x_j, y_k) \\ I(X; Y) + H(X, Y) &= H(X) + H(Y) \end{aligned}$$



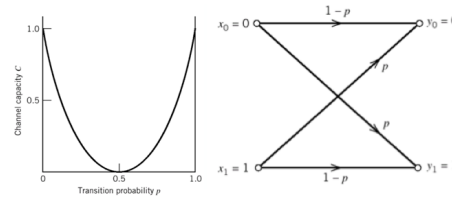
## 6.6 Channel Capacity

A channel is a statistical model with input  $X$  and output  $Y$ . Mutual information depends also on  $p(x)$ .

$$\begin{aligned} I(X; Y) &= H(Y) - H(Y|X) = \sum_{k=0}^{K-1} \sum_{j=0}^{J-1} p(x_j, y_k) \log \left( \frac{p(y_k|x_j)}{p(y_k)} \right) = \sum_{k=0}^{K-1} \sum_{j=0}^{J-1} p(y_k|x_j) p(x_j) \log \left( \frac{p(y_k|x_j)}{\sum_{j=0}^{J-1} p(y_k|x_j) p(x_j)} \right) \end{aligned}$$

**Channel capacity** is the maximum mutual information over all possible input distributions:

$$C \triangleq \max_{\{p(x_j)\}} I(X; Y) \quad C = B \log_2(1 + \text{SNR})$$



## 6.7 Channel Coding Theorem

### Shannon's Channel Coding Theorem

- Consider discrete, memoryless source emitting values in  $\mathcal{S}$
- One symbol emitted every  $T_s$  seconds - information rate  $H(S)/T_s$
- One coded symbol transmitter every  $T_c$  seconds

**Theorem** If  $H(S)/T_s < C/T_c$  there exists a channel code yielding an arbitrarily small block (message) error probability as the channel code-word length goes to infinity. For  $H(S)/T_s \geq C/T_c$  such a code does not exist.

## 6.8 Differential Entropy

Idea: Source and channels with continuous alphabets.

Differential entropy:

$$h(X) = -\int_{-\infty}^{\infty} f_X(x) \log f_X(x) dx$$

The entropy  $H(X)$  goes to  $\infty$  in the limit of  $\delta x \rightarrow 0$ . But the mutual information is well defined:

$$I(X; Y) = h(X) - h(X|Y)$$

Let  $X \sim \mathcal{N}(\mu, \sigma^2)$  be a Gaussian random variable with variance  $\sigma^2$ . Then the diff. entropy is uniquely determined by its variance.

$$h(X) = \frac{1}{2} \log(2\pi e \sigma^2)$$

## 6.9 Information Capacity Theorem

**Information capacity** is the maximum of the mutual information between input  $X_i$  and output  $Y_i$  over all distributions of  $X_i$  fulfilling the constraint:

$$C_D = \max_{\{f_X(x)\}} \{I(X_i; Y_i) : E[X_i^2] = E_s\}$$

Where  $E_s$  is interpreted as the average energy per symbol.

$$I(X_i; Y_i) = h(Y_i) - h(Y_i|X_i) = h(X_i + N_i) - h(X_i + N_i|X_i)$$

$$h(X_i + N_i) = \frac{1}{2} \log(2\pi e(E_s + \sigma^2))$$

$$h(N_i) = \frac{1}{2} \log(2\pi e \sigma^2)$$

$$C_D = \frac{1}{2} \log_2 \left( 1 + \frac{E_s}{\sigma^2} \right)$$

## 6.10 Implications of the Inf. Capacity Thm.

**ToDo:** Not yet covered in lecture

## 6.11 Colored Noise Channel

**ToDo:** Not yet covered in lecture

## 7 Data Link Layer

### 7.1 Channel Coding

The channel encoder takes the information bit sequence  $m_i$  as input and outputs coded bits  $c_j$ . Received bits are denoted by  $r_j$ . We assume a discrete memoryless channel with noise. Purpose of encoding: Change BER from problematic to acceptable for a fixed  $E_b/N_0$  or reduce required  $E_b/N_0$  for a fixed BER.

**FEC (Forward error correction)** Decoder exploits redundancy to correct errors and decide on the message bits

**Error detection** Decoder exploits redundancy to detect errors, doesn't correct them

**Block codes** have no memory in the encoder and **Convolucional codes** have memory in the encoder.

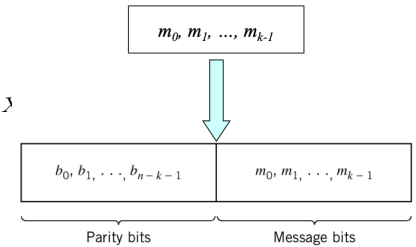
### 7.2 Linear Block Codes

$(n, k)$  linear block code takes  $k$  information bits and produces  $n$  coded bits.

- Any  $n$  code words added (mod 2) produce a third code word in the code

- The all-zero code word is part of the code

**Systematic Linear Block Code** Unaltered Message bits are extended with  $n-k$  parity bits that are linear sums of the message bits.



$$\begin{aligned} b_i &= p_{0,i}m_0 + p_{1,i}m_1 + \dots + p_{k-1,i}m_{k-1} \\ p_{j,i} &\in \{0, 1\} \end{aligned}$$

In matrix notation:

$$\begin{aligned} \mathbf{b} &= [b_0, b_1, \dots, b_{n-k-1}] \\ \mathbf{m} &= [m_0, m_1, \dots, m_{k-1}] \\ \mathbf{P} &= \begin{bmatrix} p_{00} & p_{01} & \dots & p_{0,n-k-1} \\ p_{10} & p_{11} & \dots & p_{1,n-k-1} \\ \vdots & \vdots & \ddots & \vdots \\ p_{k-1,0} & p_{k-1,1} & \dots & p_{k-1,n-k-1} \end{bmatrix} \\ \mathbf{b} &= \mathbf{mP} \end{aligned}$$

With the generator matrix  $\mathbf{G}$ :

$$\begin{aligned} \mathbf{c} &= [c_0, c_1, \dots, c_{n-1}] = [\mathbf{b}|\mathbf{m}] = \mathbf{m}[\mathbf{P}|\mathbf{I}_k] \\ \mathbf{G} &= [\mathbf{P}|\mathbf{I}_k] \quad \mathbf{c} = \mathbf{mG} \\ \mathbf{c}_i + \mathbf{c}_j &= \mathbf{m}_i\mathbf{G} + \mathbf{m}_j\mathbf{G} = (\mathbf{m}_i + \mathbf{m}_j)\mathbf{G} \end{aligned}$$

The parity check matrix  $\mathbf{H}$ :

$$\begin{aligned} \mathbf{H} &= [\mathbf{I}_{n-k}|\mathbf{P}^\top] \\ \mathbf{HG}^\top &= \mathbf{0} \quad \mathbf{cH}^\top = \mathbf{mGH}^\top = \mathbf{0} \end{aligned}$$

Multiplying a code word with the parity check matrix results in the zero vector.

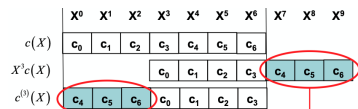
### 7.3 Cyclic Codes

Encoding and syndrome calculation with low complexity shift-registers. Practical decoding methods due to algebraic structure. **Cyclic property:** Any cyclic shift of a code word is also a code word. Description of a code via code polynomial:

$$\mathbf{c} = [c_0, c_1, \dots, c_{n-1}]$$

$$c(X) = c_0 + c_1X + c_2X^2 + \dots + c_{n-1}X^{n-1}$$

A cyclic shift is done by multiplication with  $X^j$  and modulo  $X^n+1$ . Or take the remainder of  $c(X)X^j : (X^n+1)$  with mod  $n$  arithmetic. (E.g. with  $n=7$ ,  $X^3/X^7 = X^{-4} = X^3 \mod 7$ ).



Polynomials and their order:

$m(X)$	Message polynomial	$\leq k-1$
$g(X)$	Generator polynomial	$\leq n-k$
$c(X) = m \cdot g$	Code polynomial	$\leq n-1$
$s(X) = r \mod g$	Syndrome	

If  $g(X)$  is a factor of  $X^n+1$  then the code is cyclic:

$$g(X)h(X) = X^n+1$$

$h(X)$  is the parity check polynomial, as for all  $c(X)$ :

$$\begin{aligned} c(X)h(X) &\mod (X^n+1) \\ &= m(X)g(X)h(X) \mod (X^n+1) = 0 \end{aligned}$$

**Generator Polynomial** of degree  $n-k$  that is a factor of  $X^n+1$ . Further is expanded as

$$g(X) = 1 + \sum_{i=1}^{n-k-1} g_i X^i + X^{n-k} \quad g_i \in \{0, 1\}$$

**Parity Check Polynomial**  $h(X)$

$$g(X)h(X) = X^n+1$$

**Systematic Cyclic Code Idea:** Complement the shifted message polynomial such, that the resulting code word is a multiple of  $g(X)$ . 1. Mult message by  $X^{n-k}$  2. divide by  $g(X)$  and get remainder  $b(X)/g(X)$  3. get systematic code  $\tilde{c}(X)$ .

$$m(X) = m_0 + m_1X + \dots + m_{k-1}X^{k-1}$$

$$\frac{X^{n-k}m(X)}{g(X)} = \tilde{x}(X) + \frac{b(X)}{g(X)}$$

$$\tilde{c}(X) = \tilde{m}(X)g(X) = b(X) + X^{n-k}m(X)$$

### 7.4 Minimum Distance Considerations

**Hamming distance**  $d(c_1, c_2)$  number of locations in which two code words differ.

**Hamming weight**  $w(c)$  number of non zero elements in code vector. The following hold only if linear block code:

$$d(c_1, c_2) = d(c_1 + c_2, 0) = w(c_1 + c_2)$$

**Minimum distance**  $d_{\min}$  the smallest Hamming distance between any pair of code vectors in the code. If lin. code  $d_{\min} = w_{\min}$

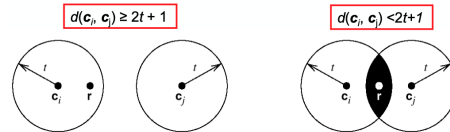
Given an  $(n, k, d_{\min})$  block code and  $t$  the number of locations where a bit toggled after transmission. **Error detection:** All error patterns with

$$t \leq d_{\min} - 1$$

can be detected. **Error correction** capability: Error patterns with weights

$$t \leq \left\lfloor \frac{1}{2}(d_{\min} - 1) \right\rfloor$$

can be corrected surely.



**Hamming bound:** For given  $d_{\min}$  and code word length  $n$ , good codes have large num. of possible code words  $2^k$ , i.e. large code rate  $r = k/n$ . The number of code words for a binary code must satisfy:

$$2^k \left( 1 + \binom{n}{1} + \dots + \binom{n}{t_0} \right) \leq 2^n \quad t_0 = \left\lfloor \frac{d_{\min} - 1}{2} \right\rfloor$$

**Perfect Code** is if a binary code  $(n, k, d_{\min})$  satisfies the hamming bound with equality.

### 7.5 Example: Hamming code

Hamming code are  $(n, k)$  lin. block codes with  $m \geq 2$  and:

$$n = 2^m - 1 \quad k = 2^m - m - 1 = n - m \quad m = n - k$$

With  $m=3$  the  $(7, 4)$  linear block code is:

$$\mathbf{G} = \left[ \begin{array}{ccc|ccc} 1 & 1 & 0 & 1 & 0 & 0 & 0 \\ 0 & 1 & 1 & 0 & 1 & 0 & 0 \\ 1 & 1 & 1 & 0 & 0 & 1 & 0 \\ 1 & 0 & 1 & 0 & 0 & 0 & 1 \end{array} \right]$$

$\underbrace{\quad\quad\quad}_{\mathbf{P}} \quad \underbrace{\quad\quad\quad}_{\mathbf{I}_k}$

**Properties of Hamming codes:**  $d_{\min} = 3$  and  $t_0 = 1$ . They satisfy the Hamming bound with equality. They are Hamming codes are single-error correcting binary perfect codes.

### 7.6 Decoding Principles

Received bit vector  $\mathbf{r} = \mathbf{c} + \mathbf{e}$  where  $\mathbf{e}$  is the error vector/pattern.

The **syndrome** is defined as the projection of  $\mathbf{r}$  onto  $\mathbf{H}$ :

$$\mathbf{s} = \mathbf{rH}^T$$

- The syndrome of  $\mathbf{r}$  depends only on the error pattern  $\mathbf{e}$
- All error patterns  $\mathbf{e}_j^i$  that differ by a code word have the same syndrome  $\mathbf{s}^i$ . Thus  $2^k$  distinct error patterns lead to the same syndrome  $\mathbf{s}^i$ .

**Syndrom of cyclic codes**

$$s(X) = r(X) \mod g(X)$$

**Standard array and coset leader** Construct a table with  $N = n - k - 1$  and  $e_i$  the most probable  $\mathbf{e}$  vectors (i.e. these with the least weight):

$c_1 = 0$	$c_2$	$\dots$	$c_{n-1}$
$e_1$	$c_2 + e_1$	$\dots$	$c_{n-1} + e_1$
$e_2$	$c_2 + e_2$	$\dots$	$c_{n-1} + e_2$
$\vdots$	$\vdots$	$\ddots$	$\vdots$
$e_N$	$c_2 + e_N$	$\dots$	$c_{n-1} + e_N$

$e_i$  are called the coset leaders. The syndrome vector points to a table entry. To obtain  $\mathbf{c}$ , XOR  $\mathbf{r}$  with  $e_i$ .

### 7.7 Maximum Likelihood Decoding

ML decoding for discrete memoryless channel is minimum Hamming distance decoding:

$$\hat{\mathbf{c}} = \arg \max_{\mathbf{c} \in C} \mathbf{P}(\mathbf{r} | \mathbf{c}) \rightarrow \hat{\mathbf{c}} = \arg \min_{\mathbf{c} \in C} wt(\mathbf{e} = \mathbf{r} + \mathbf{c})$$

Syndrome decoding is equal to ML decoding, if each coset leader has the largest probability of occurrence among all error patterns in a coset.

### 7.8 Error Probabilities

Given a BSC with transition prob.  $p$ , an  $(n, k, d)$  linear block code,  $\alpha_j$  the number of coset leaders with weight  $j$ , the error probability is given by:

$$P_e = 1 - \underbrace{\sum_{j=0}^n \alpha_j p^j (1-p)^{n-j}}_{\mathbf{P}(\text{error pat} = \text{coset leader})} \quad \alpha_0 = 1$$

**Error Detection:**  $2^n - 2^k$  error patterns are detectable.  $2^k - 1$  undetectable error patterns (zero syndrome). Probability  $P_u$  of an undetectable error in BSC with  $p$ ,  $(n, k, d)$  lin. block code,  $w_j$  number of code words with weight  $j$ ,  $P_u$  is the prob. that an error pattern itself is a code word:

$$P_u = \sum_{j=1}^n w_j p^j (1-p)^{n-j}$$

## 8 Convolutional Codes

### 8.1 Convolutional Encoder

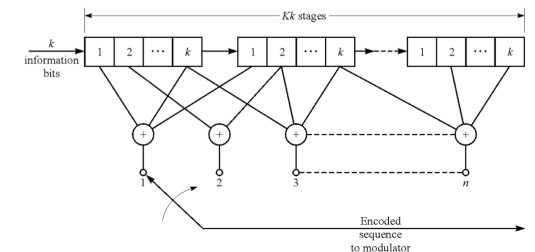
Consists of a shift register that shifts  $k$ -bit words at a time. Is a linear code.  $n$  Generators used to generate encoded sequence. For  $Nk$  input bits with known header and trailer ensures defined SR content and defines a  $(n(N+K-1), Nk)$  lin. block code.

$k$  stage length, mostly  $k=1$

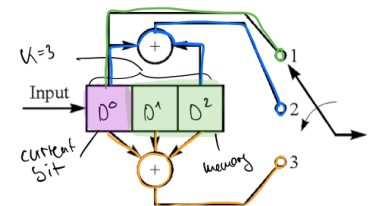
$K$  Constraint length: Number of  $k$ -bit stages

$k/n$  Approximate code length

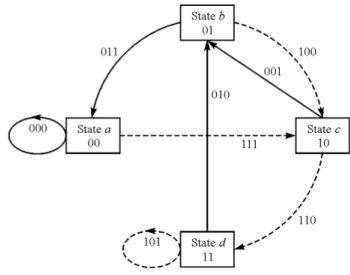
$n$  Number of generators



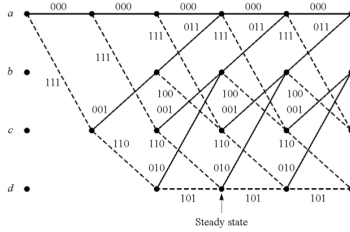
**Example**  $(n=3, k=1, K=3)$   $(3, 1, 3)$  conv. encoder. Rate  $R_c = k/n = 1/3$ . Systematic: Coded bit 1 is information bit.  $g_2 = [101] = 1 + 0 \cdot D + 1 \cdot D^2$



Corresponding **state diagram**. Solid arrows represent new message bit 0, dashed 1. Code word is denoted next to edges.



The **trellis diagram** denotes the states vertically and the progression of input message bits horizontally. Again solid (dashed) edges = message 0 (1) and code word aside edge.



Hamming weight: Weight of concatenated code-words of path through trellis  
Free Hamming distance: Minimum weight of code-words.

**Transfer function** can be derived from state diagram. Transitions labeled with weight  $D^m$  where  $m$  is the Hamming weight of the associated coded bits. For each state there is an equilibrium condition:

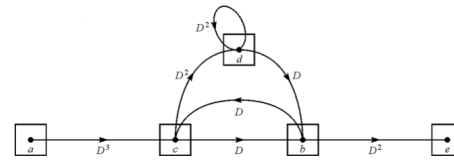
$$\begin{aligned} b : & DX_c + DX_d - X_b = 0 \\ c : & X_a D^3 + X_b D - X_c = 0 \\ d : & D^2 X_d + D^2 X_c - X_d = 0 \\ e : & D^2 X_b = X_e \end{aligned}$$

In Matrix form:

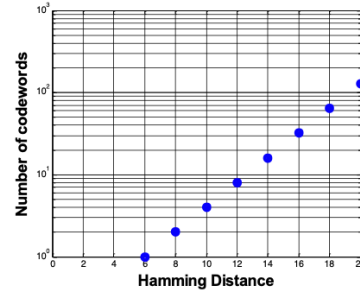
$$\begin{bmatrix} D & D & -1 & 0 \\ D^2 - 1 & D^2 & 0 & 0 \\ 0 & -1 & D & D^3 \\ 0 & 0 & D^2 & 0 \end{bmatrix} \begin{bmatrix} X_d \\ X_c \\ X_b \\ X_a \end{bmatrix} = \begin{bmatrix} 0 \\ 0 \\ 0 \\ X_e \end{bmatrix}$$

After gaussian elimination yields the transfer function

$$\begin{aligned} T(D) &= \frac{X_e}{X_a} = \frac{D^6}{1 - 2D^2} \\ &= D^6 + 2D^8 + 4D^{10} + 8D^{12} + L \end{aligned}$$



Polynomial division results in **distance distribution**.  $8D^{12}$ : 8 codewords with distance 12.



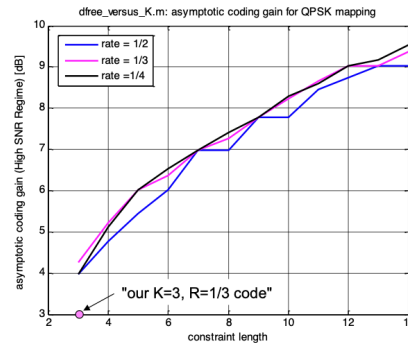
**Error Probability:** Pairwise prob. of error between two codewords with Hamming distance  $d_H$ :

$$P_2(d_H) = Q\left(\sqrt{\frac{2E_b}{N_0} R_c d_H}\right) \quad E_c = E_b R_c$$

$E_c$  Energy per coded bit

**Coding Gain** is asymptotic and determines performance in the high SNR regime.

$$G_a = 10 \log_{10}(R_c d_{H, \text{free}})$$



## 8.2 Viterbi Decoder

**Conventions**

$m = K - 1$  memory depth  
( $n, k, m$ ) codes comprise  $2^{mk}$  states  
 $u, v$  input/coded bits  
 $\mathbf{v}^{(i)}$  Encoded sequence

$\mathbf{r}$  Received sequence  
 $p(\mathbf{r}|\mathbf{v}^{(i)})$  ML metric of trial path

$$\begin{aligned} \mathbf{v}^{(i)} &= [\mathbf{v}_0^{(i)} \quad \dots \quad \mathbf{v}_{h+m-1}^{(i)}] \\ \mathbf{r} &= [\mathbf{r}_0 \quad \dots \quad \mathbf{r}_{h+m-1}] \\ p(\mathbf{r}|\mathbf{v}^{(i)}) &= \prod_k p(\mathbf{r}_k|\mathbf{v}_k^{(i)}) \end{aligned}$$

$M(\mathbf{r}|\mathbf{v}^{(i)})$  branch metric

$$\begin{aligned} M(\mathbf{r}|\mathbf{v}^{(i)}) &\triangleq \log p(\mathbf{r}|\mathbf{v}^{(i)}) = \sum_k \log p(\mathbf{r}_k|\mathbf{v}_k^{(i)}) \\ &= \sum_k M(\mathbf{r}_k|\mathbf{v}_k^{(i)}) \end{aligned}$$

**BSC:** Special case for binary sym. channel

$$\begin{aligned} \mathbf{P}(r_{l,j} = 1 | v_{l,j} = 1) &= \mathbf{P}(r_{l,j} = 0 | v_{l,j} = 0) = 1 - p \\ \mathbf{P}(r_{l,j} = 0 | v_{l,j} = 1) &= \mathbf{P}(r_{l,j} = 1 | v_{l,j} = 0) = p \end{aligned}$$

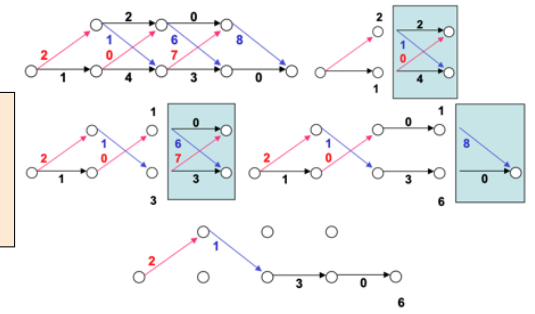
**AWGN:** Special case for AWGN channel

$$\log \mathbf{P}(r_{l,j} = a | v_{l,j} = b) = -|a - b|^2$$

## Algorithm

- Begin at time  $t = m$ , compute partial metric for single path entering each state. Store path (survivor) and its metric for each state.
- Increase  $t$  by 1  
**Branch metric:** Compute branch metric for all  $2^k$  paths entering a state  
**Add:** Compute the partial path metrics by adding the branch metric entering that state to the metric of the connecting survivor at the previous time  
**Compare** the partial metrics for each state for all  $2^k$  paths entering that state  
**Select** the path with best metric (survivor), store it with its metric and eliminate all other paths
- Iterate until  $t = h + m + 1$

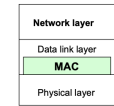
**Example**



## 9 Multiple Access Protocols

### 9.1 Basics of Channel Access

In broadcast networks, one channel is shared by multiple users, therefore coordination is required. The medium access control (MAC) sublayer controls access to the shared channel.



**Static channel allocation:** Frequency- (FDM), Time- (TDM), Code- (CDM), Space-Division (SDM) multiplexing.

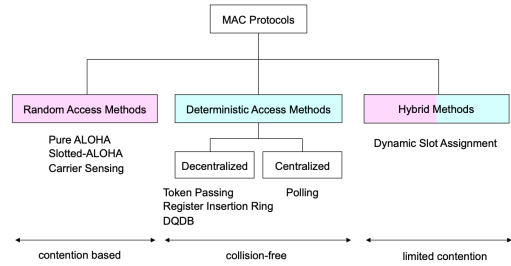
**Dynamic channel allocation:** Time variable traffic from different sources. Collided frames will be retransmitted and can be avoided using suitable coordination among stations.

Assumptions:

- Station Model:  $N$  independent stations generating traffic.
- A single channel is available for all transmissions
- Collision: If two frames overlap in time the resulting signal is distorted.
- Cont. Time Alloc: Tx can be performed in any time instant
- Slotted Time: Time is divided into discrete slots. Tx begins at start of slot.
- Carrier Sense: Stations listen to channel before Tx



## 9.2 MAC Protocol Classification



## 9.3 ALOHA Family Protocols

Data can be sent at any time. If collision, a random waiting time is passed and the data is retransmitted.

$D$  transfer time / frame length [s]

$g$  offered load [frames/s]

$G$  offered load

$S$  throughput [frames] per frame duration

$P_0$  prob. of successful transmission

$$G = gD \quad S = GP_0$$

### Slotted ALOHA

$$P_0 = e^{-G} \quad S = Ge^{-G} \leq \frac{1}{e} \quad \text{eqty with } G = 1$$

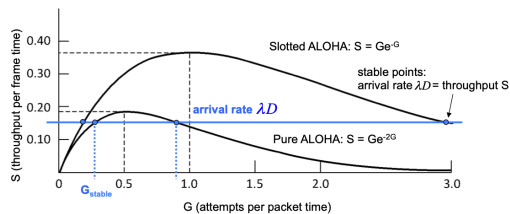
### Unslotted (Pure) ALOHA

$$P_0 = e^{-2G} \quad S = Ge^{-2G} \leq \frac{1}{2e} \quad \text{eqty with } G = 0.5$$

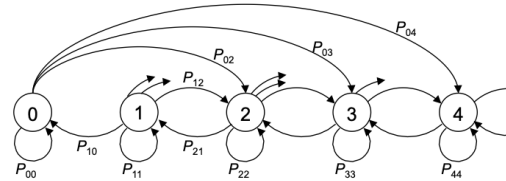
Prob. that  $k$  frames are transmitted during time  $T$  with large number of stations is poisson with  $\lambda = gT, m_k = \sigma_k^2 = gT$ :

$$P_0(k|T) = \frac{(gT)^k e^{-gT}}{k!} \quad T = \begin{cases} D, \text{ slotted,} \\ 2D, \text{ unslotted.} \end{cases}$$

$$P_0(k=0|T) = e^{-gT} \quad P_0(k=1|T) = gTe^{-gT}$$



**Markov Chain Model** For low number of stations  $N < 10$  a markov chain can be used for calculating probabilities. A station is backlogged if it encountered a collision during Tx and has to retransmit. The MC state represents the number of backlogged stations.



$$P = \begin{bmatrix} P_{0,0} & \dots & P_{0,m} \\ \vdots & \ddots & \vdots \\ P_{m,0} & \dots & P_{m,m} \end{bmatrix} \quad \mathbf{p}_{j+1} = P^T \mathbf{p}_j$$

$$j \rightarrow \infty \quad \mathbf{p}_{j+1} = \mathbf{p}_j = P^T \mathbf{p}_j$$

Where  $j \rightarrow \infty$  denotes the steady state that can be calculated using a eigenvalue problem.

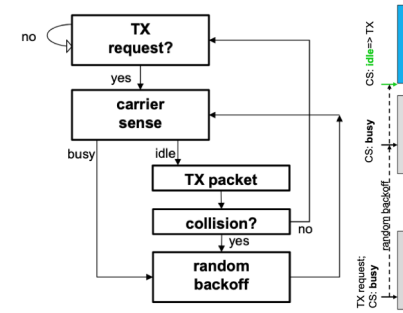
nr. of new frames	nr. of retransmitted frames	contribution to transition	Probability
0	0	$\circ \rightarrow \circ$	$P_s(0,n) \cdot P_s(0,n)$
1	0	$\circ \rightarrow \circ$	$P_s(1,n) \cdot P_s(0,n)$
0	1	$\circ \rightarrow \circ$	$P_s(0,n) \cdot P_c(1,n)$
1	1	$\circ \rightarrow \circ$	$P_s(1,n) \cdot P_c(1,n)$

nr. of new frames	nr. of retransmitted frames	contribution to transition	Probability
$i_1 > 1$	0	$\circ \rightarrow \circ$	$P_s(i,n) \cdot P_s(0,n)$
0	$> 1$	$\circ \rightarrow \circ$	$P_s(0,n) \cdot (1 - P_c(0,n) - P_c(1,n))$
$i_1 > 1$	1	$\circ \rightarrow \circ$	$P_s(i,n) \cdot P_c(1,n)$
1	$> 1$	$\circ \rightarrow \circ$	$P_s(1,n) \cdot (1 - P_c(0,n) - P_c(1,n))$
$i_1 > 1$	$> 1$	$\circ \rightarrow \circ$	$P_s(i,n) \cdot (1 - P_c(0,n) - P_c(1,n))$

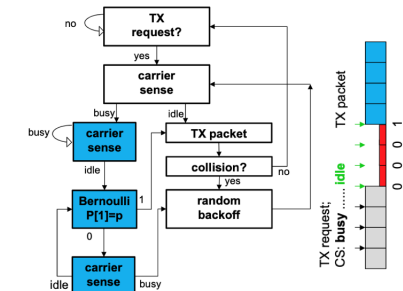
## 9.4 CSMA Carrier Sense Multiple Access

Listen for a ongoing transmission and act accordingly.

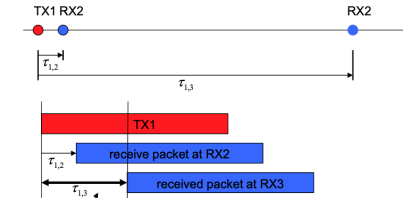
**Nonpersistent** Wait random time after check channel again and loop. May not need to continually sense the channel.



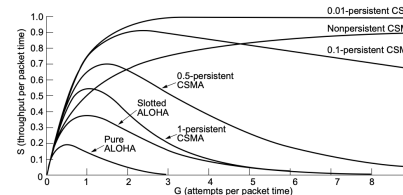
**p-persistent** Wait for idle, Tx with probability  $p$ , repeat until Tx or other Tx then start again. 1-persistent has  $p = 1$  and sends immediately as soon as channel is free.



### Vulnerable Period



### Channel Utilization

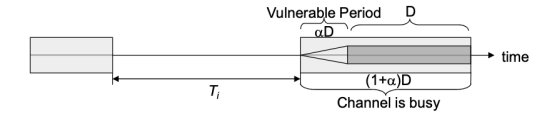


**Throughput** Many stations  $m \rightarrow \infty$ , transmission with fixed duration  $D$ , channel access Poisson distributed with rate  $g$ . Normalized offered load  $G = gD$ . Possible channel states: Idle, successful Tx, collision.

$\tau_{\max} = \alpha D$  Worst case vulnerable period

$\tau_{\max}$  Max. propagation delay

$P(k|T)$  Prob. of  $k$  frames in  $T$   
 $D + \tau_{\max} = (1 + \alpha)D$  duration of channel access without collision



$$p_T(T_i) = g \exp(-gT_i) \quad E[T] = \int_0^\infty t p_T(t) dt = 1/g$$

Prob. that no other packet is Tx in vulnerable period:

$$P(\text{success}) = e^{-\alpha G}$$

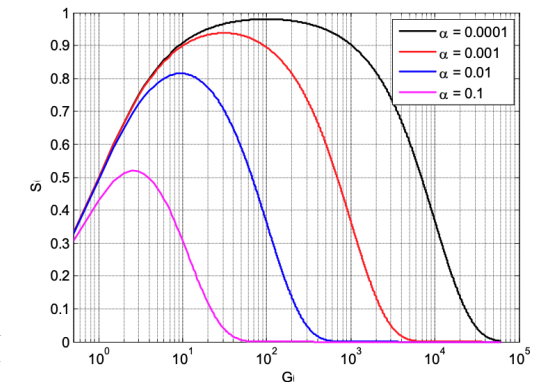
Mean duration of a Tx cycle:  $T + (1 + \alpha)D$

$$s \approx \frac{P(\text{success})}{T + (1 + \alpha)D} = \frac{e^{-\alpha G}}{T + (1 + \alpha)D}$$

Normalized throughput per packet duration  $D$

$$S = sD = \frac{e^{-\alpha G}}{\frac{1}{G} + 1 + \alpha}$$

$$\lim_{\alpha \rightarrow 0} S = \frac{G}{G+1} \quad \lim_{G \rightarrow \infty} S = 0$$



## 9.5 CSMA/CD collision detect

Detect a collision and stop Tx.



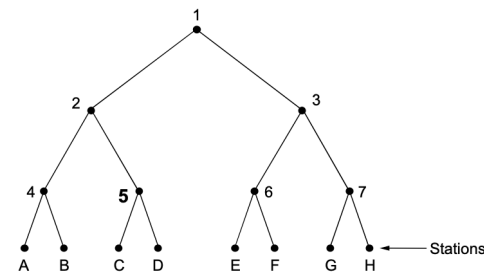
### 9.6 Binary exponential backoff

Based on CSMA/CD, three channel states: idle, contention, success.

- After collision, time is divided into discrete slots: length of each slot is equal to worst-case round-trip propagation time  $2\tau$
- After first collision, each station waits either 0 or 1 slot times before trying again
- After each further collision the backoff window is doubled (up to max 1024)
- In general after  $i$  collisions, a random number between 0 and  $2^i - 1$  is chosen, and that number of slots is skipped
- after 16 collisions, the controller reports failure to higher layer

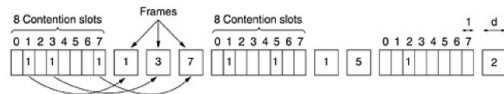
- If one acquires channel, the slot following the frame is reserved for those stations under node 3. If collision again, go down to 4

The heavier the load, the farther down the tree the search should begin



### 9.7 Collision-Free Protocols

**Bit-Map Protocol:** Contention slots and frames. Each station wanting to send, transmits 1 during contention slots. Frames can then be sent in order of address. Addresses can be rotated to prevent starvation.



### 9.8 Limited-Contention Protocols

Idea: Use contention at low load (to provide low delay) but use a collision-free technique at high load (to provide good channel efficiency).

Assumptions:

- We allow  $k$  stations to contend for channel access
- each station has a probability  $p$  of transmitting during each slot

Probability of successful transmission is  $P(\text{success}) = kp(1-p)^{k-1}$ . Optimum value for  $k = 1/p$ .

$$P(\text{Success with opt. } k | p) = (1-p)^{1/p-1}$$

$$P(\text{Success with opt. } p | k) = \left(\frac{k-1}{k}\right)^{k-1}$$

### Adaptive Tree Walk Protocol

- First, all stations are allowed to tx
- If collision, during slot 1 only stations under node 2 may compete

## 10 Math

### 10.1 General

$$\cos(a)\cos(b) + \sin(a)\sin(b) = \cos(a-b)$$

$$\cos(a+b)\cos(a-b) = \frac{1}{2} [\cos(2a) + \cos(2b)]$$

$$\cos(a)\cos(b) = \frac{1}{2} (\cos(a-b) + \cos(a+b))$$

$$\sin(a)\sin(b) = \frac{1}{2} (\cos(a-b) - \cos(a+b))$$

$$\cos(a)\sin(b) = \frac{1}{2} (\sin(a+b) - \sin(a-b))$$

$$\text{sinc}(x) = \frac{\sin(\pi x)}{\pi x}$$

$$\sin(x) = \frac{e^{ix} - e^{-ix}}{2i} \quad \cos(x) = \frac{e^{ix} + e^{-ix}}{2}$$

### 10.2 Fourier Transform

Source: Haykin, Communication systems, 4th ed.

$$\text{rect}\left(\frac{t}{T}\right) \longleftrightarrow T \text{sinc}(fT)$$

$$\text{sinc}(2Wt) \longleftrightarrow \frac{1}{2W} \text{rect}\left(\frac{f}{2W}\right)$$

$$\exp(-at)u(t), a > 0 \longleftrightarrow \frac{1}{a + j2\pi f}$$

$$\exp(-a|t|), a > 0 \longleftrightarrow \frac{2a}{a^2 + (2\pi f)^2}$$

$$\exp(-\pi t^2) \longleftrightarrow \exp(-\pi f^2)$$

$$\begin{cases} 1 - \frac{|t|}{T}, & |t| < T \\ 0, & |t| \geq T \end{cases} \longleftrightarrow T \text{sinc}^2(fT)$$

$$\delta(t) \longleftrightarrow 1$$

$$1 \longleftrightarrow \delta(f)$$

$$\delta(t - t_0) \longleftrightarrow \exp(-j2\pi f t_0)$$

$$\exp(j2\pi f_c t) \longleftrightarrow \delta(f - f_c)$$

$$\cos(2\pi f_c t) \longleftrightarrow \frac{1}{2} [\delta(f - f_c) + \delta(f + f_c)]$$

$$\sin(2\pi f_c t) \longleftrightarrow \frac{1}{2i} [\delta(f - f_c) - \delta(f + f_c)]$$

$$\text{sgn}(t) \longleftrightarrow \frac{1}{j\pi f}$$

$$\frac{1}{\pi t} \longleftrightarrow -j \text{sgn}(f)$$

$$u(t) \longleftrightarrow \frac{1}{2} \delta(f) + \frac{1}{j2\pi f}$$

$$\sum_{i=-\infty}^{\infty} \delta(t - iT_0) \longleftrightarrow \frac{1}{T_0} \sum_{n=-\infty}^{\infty} \delta\left(f - \frac{n}{T_0}\right)$$

$u(t)$  unit step function

$\delta(t)$  delta function

$\text{rect}(t)$  rectangular function of unit amplitude and unit duration centered on the origin

$\text{sgn}(t)$  signum function

$\text{sinc}(t)$  sinc function

Relations

$$\alpha f(t) + \beta g(t) \longleftrightarrow \alpha F(f) + \beta G(f)$$

$$f^*(t) \longleftrightarrow F^*(-f)$$

$$f(at) \longleftrightarrow \frac{1}{|a|} F\left(\frac{f}{a}\right)$$

$$f(t-a) \longleftrightarrow e^{-j2\pi f a} F(f)$$

$$e^{j2\pi f_0 t} f(t) \longleftrightarrow F(f - f_0)$$

$$f^{(n)} \longleftrightarrow (j2\pi f)^n F(f)$$

$$t^n f(t) \longleftrightarrow j^n F^{(n)}(f)$$

$$\int_{-\infty}^t x(\tau) d\tau \longleftrightarrow \frac{1}{j2\pi f} F(f) + \pi F(0) \delta(f)$$

$$\frac{1}{t} x(t) + \pi x(0) \delta(t) \longleftrightarrow \int_{-\infty}^f X(s) ds$$

$$(f * g)(t) \longleftrightarrow F(f) \cdot G(f)$$

$$f(t) \cdot g(t) \longleftrightarrow \frac{1}{2\pi} F(f) * G(f)$$

$f^{(n)}$   $n^{\text{th}}$  derivation

$f^*$  complex conjugate

### 10.3 Sums

$$\sum_{k=0}^n q^k k = \frac{nq^{n+2} - (n+1)q^{n+1} + q}{(q-1)^2}$$

$$\sum_{k=0}^n k = \frac{n(n+1)}{2}$$

$$\sum_{k=0}^n k^2 = \frac{n(n+1)(2n+1)}{6}$$

$$\sum_{k=0}^n k^3 = \frac{n^2(n+1)^2}{4}$$

$$\sum_{k=0}^{\infty} q^k = \frac{1}{1-q}$$

$$\sum_{k=1}^{\infty} q^k = \frac{q}{1-q}$$

### 10.4 Probability

$$\mathbf{P}(X > x) = \frac{1}{2} \text{erfc}\left(\frac{x}{\sqrt{2}\sigma_X}\right)$$

$$\text{erf}(x) = \frac{1}{\sqrt{\pi}} \int_{-x}^x e^{-t^2} dt = \frac{2}{\sqrt{\pi}} \int_0^x e^{-t^2} dt$$

$$\text{erfc}(x) = 1 - \text{erf } x = 2Q(\sqrt{2}x) = \frac{2}{\sqrt{\pi}} \int_x^{\infty} e^{-t^2} dt$$

$$Q(x) = \frac{1}{\sqrt{2\pi}} \int_x^{\infty} \exp(-y^2/2) dy \leq \exp(-x^2/2)$$

$$Q(x) = \frac{1}{2} \text{erfc}\left(\frac{x}{\sqrt{2}}\right)$$

