Projects in Wireless Communication Lecture 1

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- Introduction to the course
- Basics of digital communications
- Discrete-time implementations
- Carrier transmission



Introduction

Lecturer and course responsible: Fredrik Rusek, E:2377 5 scheduled lectures

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Ultimate goal for PWC:

Two computers should communicate via speaker/microphones

We aim at a file-transfer and/or a conversation via the keyboards Some form of advanced system should be implemented, e.g. MIMO, OFDM, Turbo coding etc

The projects should be performed in groups of TWO students (HARD LIMIT)



In the first part of PWC we only work in software. For a passing grade you must solve three tasks:

1. A digital baseband BPSK system should be implemented in C++ and its performance should be measured and verified against theoretical results

$$P_e = \mathcal{Q}\left(\sqrt{d_{\min}^2 \frac{E_b}{N_0}}\right)$$

2. Later in PWC you will encounter physical passband signals at the input of the microphone. In the first part, we will provide each group with one such signal; the bits carried by the signals correspond to the ASCII code of a secret password. If you can decode the signals and provide me with the password, you have passed task 2.

3. Same as 2 but with OFDM transmission and convolutional code.





Assume that you receive the following noisy signal

You must remove the noise...





Assume that you receive the following noisy signal

You must remove the noise...Done!





Assume that you receive the following noisy signal

You must remove the noise...Done! Decode the bits:





Assume that you receive the following noisy signal

You must remove the noise...Done! Decode the bits: 1 1 1 0 0 0 0 0 1 1 0 1 0 0 1 1 1 0 1.....





Assume that you receive the following noisy signal

You must remove the noise...Done! Decode the bits: 1 1 1 0 0 0 0 0 1 1 0 1 0 0 1 1 1 0 1.... Convert to ASCII:





Assume that you receive the following noisy signal

You must remove the noise...Done! Decode the bits: 1 1 1 0 0 0 0 0 1 1 0 1 0 0 1 1 1 0 1.... Convert to ASCII: You have passed PWC1, congratulations.....



Formal descriptions of the tasks can be found online.



This is a recall of baseband digital communications....

We need to transmit a bit sequence $\{u_k\} = 0111010....$ Map to symbols $\{a_k\}$

BPSK:
$$a_k = \begin{cases} 1, u_k = 0 \\ -1, u_k = 1 \end{cases}$$

QPSK: $a_k = \begin{cases} 1, u_{2k}u_{2k+1} = 00 \\ i, u_{2k}u_{2k+1} = 01 \\ -1, u_{2k}u_{2k+1} = 10 \\ -i, u_{2k}u_{2k+1} = 11 \end{cases}$



Each symbol is carried by a base pulse p(t) of length T, e.g. the half-cycle sinus





So the transmission of bits 0 1 0 0 0 0 1 generates the pulse train y(t)



Mathematically we have

$$y(t) = \sum_{k} a_k p(t - kT_s)$$

Note that T_s is the symbol time while T is the duration of the base pulse p(t).



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In this example we have $T = T_s/2$



The channel model assumed in this review is a pure AWGN channel



Where the noise n(t) satisfies $\mathcal{E}\{n^*(t)n(t+\tau)\}=\delta(\tau)N_0/2$; such a noise process must have power spectral density



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Consider the power of the process

$$P = \int R(f) \mathrm{d}f$$



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Consider the power of the process

$$P = \int R(f) \mathrm{d}f$$

n(t) has infinite power!

Thus, not possible to show an example of WGN



Explanation: Every signal we ever see in reality has been filtered by some low-pass filter.



Mathematically, in what way should the receiver process the received signal r(t).

In other words

 $\hat{\mathbf{a}} = \dots ?$



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Maximum-likelihood detection is the answer!

 $\hat{\mathbf{a}} = \arg\max_{\mathbf{a}} Prob\{r(t)|\mathbf{a}\}$



Mathematically, in what way should the receiver process the received signal r(t).

Maximum-likelihood is equivalent to minimum Euclidean distance detection

$$\hat{\mathbf{a}} = \arg\min_{\mathbf{a}} \int_{-\infty}^{\infty} |r(t) - \sum_{k} a_{k} p(t - kT_{s})|^{2} dt$$



To decode the (complex valued) signal r(t), we pass r(t) through a matched filter z(t)

$$z(t) = p(-t)$$

For symmetric pulses p(t), we get

$$z(t) = p(t)$$

Let

$$\begin{aligned} x(t) &= r(t) \star p(t) \\ &= \sum_k a_k g(t - kT_s) + \eta(t) \end{aligned}$$

where $\eta(t)$ is $n(t)\star p(t)$ and $g(t)=p(t)\star z(t).$ Take samples every T_s seconds: $x_k=x(kT_s).$ Then

$$x_k = E_p a_k + \eta_k$$

where η_k is a complex Gaussian random variable with variance $E_p N_0$, that is $E_p N_0/2$ per dimension!



Energy computations and error probability:

The energy per transmitted symbol
$$E_s$$
 is given by: $E_s = \underbrace{\int p^2(t) dt}_{E_p}$ while the energy

per transmitted bit is

$$E_b = \begin{cases} E_s, \text{ BPSK} \\ E_s/2, \text{ QPSK} \end{cases}$$

The physical minimum Euclidean distance is

$$D_{\min}^2 = \begin{cases} 4E_p, & \text{BPSK} \\ 2E_p, & \text{QPSK} \end{cases}$$

In both cases we end up with a normalized distance $d_{\min}^2 = 2$. The error probability is given by

$$P_e \approx \mathcal{Q}\left(\sqrt{2\frac{E_b}{N_0}}\right)$$



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In a computer-based package such as Matlab or C/C++, we cannot represent the signals y(t) as continuous time signals. Hence we must work with sampled versions.

Let f_s be the sample rate in samples/second and N be the number of samples per symbol.

In PWC2, $f_s = 44100$ samples/second

We get that $T_s = \frac{N}{f_s}$

The symbol rate becomes

$$R_s = \frac{f_s}{N}$$



We must sample the base pulse p(t).







We must sample the base pulse p(t). Assume a sample interval of T_s/N seconds



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We must sample the base pulse $p(t). \ {\rm N+1}$ samples per symbol implies sample interval of T_s/N seconds



This is wrong!



Explanation: Plot two consecutive pulses.



There should only be one point.



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Correct sampling!



Represent the samples in a vector

$$p = [0 \ 0.159 \ .309 \ ...]$$



A sampled transmission signal of + - + + - +



Slightly harder mathematical representation. Let $\{b_k\}$ be a zero-padded version of $\{a_k\}$

$$\boldsymbol{b} = \begin{bmatrix} a_1 & \underbrace{00 \dots 0}_{N-1} & a_2 & \underbrace{00 \dots 0}_{N-1} & a_3 & \underbrace{00 \dots 0}_{N-1} & a_4 \dots \end{bmatrix}$$

Then,

$$y_k = \sum_{\ell} b_{\ell} p_{k-\ell}$$
 or simply $\boldsymbol{y} = \boldsymbol{b} \star \boldsymbol{p}$



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Convolutions in discrete-time:

A convolution of x(t) and y(t) in continuous time is carried out as

$$\int x(\tau)y(t-\tau)\mathrm{d}\tau\tag{1}$$

Let x and y be sampled version of x(t) and y(t); the sampling rate is f_s . The discrete time version of (1) is

$$\frac{1}{f_s} \sum_{\ell} x_{\ell} y_{k-\ell}$$

The discrete time convolution must be scaled by the sampling rate!. $1/f_s$ works as $d\tau$ in (1).

The energy of the pulse p(t) must be approximated as

$$E_p = \frac{1}{f_s} \sum_k p_k^2$$



Matched filters in discrete-time:

The pulse train p should be filtered by a discrete-time matched filter. For symmetric pulses, we can take this mathched filter as z = p where p includes the last sample!, i.e. the length of p is N+1. (This is however not crucial.) Then the output of the matched filter is (N = 20)



The number of samples in y is $N \times$ number of symbols and the length of the filter output is $N + N \times$ number of symbols. The peak occurs at samples 1 + kN, k = 1, 2, 3.



If there is a guard band $(T < T_s)$, then the pulse is not symmetric and we can not take z = p. We must then use

$$z_k = p_{N+2-k}, \quad k = 1...N+1$$

It is still true that the peaks occur at samples 1 + kN, k = 1, 2, 3...



Implementation of discrete-time AWGN:

Until now we have constructed a modulation signal y in discrete time. We now seek a noise vector n to be added to y that represents continuous time AWGN (that has inf power).

We have that both the real and the imaginary parts of the samples of

 $\eta(t) = n(t) \star z(t)$

are zero-mean and have variance $E_p N_0/2$.

In discrete-time, a sample of the filtered noise process is given by

$$\eta_k = \frac{1}{f_s} \sum_{\ell} n_\ell z_{k-\ell}$$

Assume that the variance of each n_k is σ_n^2 . From probability theory it follows that η_k has variance $\sigma_n^2 \sum z_k^2 / f_s^2$.

Since

$$\sigma_n^2 \sum_k z_k^2 / f_s^2 = E_p \frac{N_0}{2}$$





we get that

$$\sigma^2 = E_p \frac{N_0}{2} \frac{f_s^2}{\sum_k z_k^2} = \frac{N_0}{2} f_s$$

Thus, The sampling rate affects the variance of the discrete time representation of continuous AWGN



The transmitted signal is $y(t) = \sum_k a_k h(t - kT)$. What is the bandwidth? More generally, what is its Fourier transform?



Cable 2.3 Properties of the Fourier transform	
 Linearity Inverse Translation (time shift) Modulation (frequency shift) 	$ax_{1}(t) + bx_{2}(t) \leftrightarrow aX_{1}(f) + bX_{2}(f)$ $x(t) = \int_{-\infty}^{\infty} X(f) e^{j\omega t} df$ $x(t - t_{0}) \leftrightarrow X(f) e^{-j\omega t_{0}}$ $x(t) e^{j\omega_{0}t} \leftrightarrow X(f - f_{0})$ $x(t) \cos \omega_{0}t \leftrightarrow \frac{1}{2}X(f + f_{0}) + \frac{1}{2}X(f - f_{0})$
5. Time scaling	$x(at) \Leftrightarrow \frac{1}{ a } X(f/a)$
6. Differentiation in time	$\frac{d}{dt} x(t) \leftrightarrow j\omega X(f)$
Differentiation in frequency	$tx(t) \Leftrightarrow -\frac{1}{j2\pi} \frac{d}{df} X(f)$
8. Integration in time	$\int_{-\infty}^{t} x(\tau) d\tau \leftrightarrow \frac{1}{j\omega} X(f)$
 9. Duality 10. Conjugate functions 11. Convolution in time 12. Multiplication in time 13. Parseval's formulas 	$X(t) \Leftrightarrow x(-f)$ $x^{*}(t) \leftrightarrow X^{*}(-f)$ $x_{1}(t) * x_{2}(t) \Leftrightarrow X_{1}(f)X_{2}(f)$ $x_{1}(t)x_{2}(t) \leftrightarrow X_{1}(f) * X_{2}(f)$ $\int_{-\infty}^{\infty} x_{1}(t)x_{2}^{*}(t) dt = \int_{-\infty}^{\infty} X_{1}(f)X_{2}^{*}(f) df$ or, when $x_{1}(t) = x_{2}(t)$, $\int_{-\infty}^{\infty} x(t) ^{2} dt = \int_{-\infty}^{\infty} X(f) ^{2} df$



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The carrier modulated signal is $y_m(t) = y(t) \cos(2\pi t f_c)$ But bandwidth gets twice as large!



Where did the energy go?

Basic Fourier relations:

$$\cos(2\pi f_c t)h(t) \longleftrightarrow \frac{1}{2}H(f - f_c) + \frac{1}{2}H(f + f_c)$$
$$\sin(2\pi f_c t)h(t) \longleftrightarrow \frac{i}{2}H(f - f_c) - \frac{i}{2}H(f + f_c)$$



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What about the increased bandwidth?



Assume two independent baseband transmissions





Assume two independent baseband transmissions After modulation with $\cos(2\pi t f_c)$ and $\sin(2\pi t f_c)$ we get





Assume two independent baseband transmissions After demodulation with $\cos(2\pi t f_c)$ we get



The red spectras cancel out, thus, we can detect the blue independently from the red Similar for demodulation with $\sin(2\pi t f_c)$



The block diagram of the transmitter is



$$y(t) = y_I(t)\cos(2\pi f_c t) - y_Q(t)\sin(2\pi f_c t)$$



The block diagram of the receiver is



The in-phase and the quadrature components can be independently detected! The LPF (low pass filters) can be taken as a matched filter to h(t)



The signals at both rails are baseband signals, and conventional processing follows: matched filter \rightarrow sampling every T_s second \rightarrow decision unit





What is a complex-valued symbol 1 + i?

In QPSK, we transmit complex valued symbols. In one symbol interval, we have

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Real part goes here and imaginary here



We can alternatively express the signal $\boldsymbol{y}(t)$ as

$$y(t) = y_I(t)\cos(2\pi f_c t) - y_Q(t)\sin(2\pi f_c t)$$

= $e(t)\cos(2\pi f_c t + \theta(t))$

where e(t) is the envelope and $\theta(t)$ is the phase

For QPSK, $e(t) = \sqrt{2}h(t)$ and $\theta(t) \in \{\pi/4, 3\pi/4, 5\pi/4, 7\pi/4\}$

We can further manipulate y(t) into

$$y(t) = \operatorname{Re}\{(y_I(t) + iy_Q(t))e^{2\pi f_c t}\}$$

= $\operatorname{Re}\{\tilde{y}(t)e^{i2\pi f_c t}\}$

where

$$\tilde{y}(t) = y_I(t) + i y_Q(t)$$



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We can either do this as

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or as

$$y(t) = \sqrt{2}h(t)\cos(2\pi f_c t + 7\pi/4)$$



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or as

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or as

$$y(t) = \operatorname{Re}\{(1-i)h(t)e^{2\pi f_c t}\}$$



In the last representation, we can change the receiver processing into



