Matched Filter in octave

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Procedure followed in simulations:

first the pulse vector p was created in octave with sufficient data points and supplied sampling frequency fs. the point at which the point p falls below 10^{-2} , supplied to the program as "tolerance" was found out and the sequence $\langle p \rangle$ was truncated (as p_trunc).

length of the sequence p_trunc for a sampling frequency of 10 Hz and tolerance of 0.01 error in amplitude was found to be 27.

10000 symbols were generated randomly using rand function as

$$sk = -1 + 2 * (rand(1, k) > 0.5);$$

The symbols were modulated with the truncated pulse $p_t runc$ and passed through the filter h[n], impulse invariance was used for sampling

$$i.e. h_d[n] = T_s * h_c(nT_s)$$

$$\left(\sum_{k} s_{k}.p_{trunc}[n-kN]\right) * h_{trunc}[n]$$

h[n] was truncated to an amplitude error tolerance of 10^{-3} . AWGN noise was added to the modulated signal using randn function

$$tx_sig = signal + f(\sigma). * randn();$$

where randn follows a normal distribution; matched filter was constructed as q[-n] where

area filter was constitueted as
$$g[n]$$
 where

the recieved message was convolved with the matched filter and sampled at intervals of L where L is the length of p_{trunc} and bit error rate was calculated and plotted

 $g = p_{trunc} * h_{trunc}$

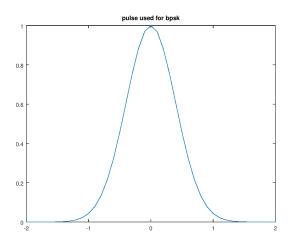


Figure 1.1: truncated pulse used for signalling

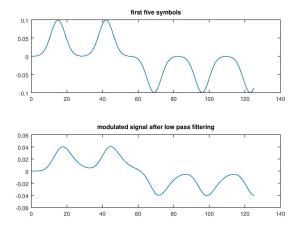


Figure 1.2: transmitted signal (noise is not introduced)

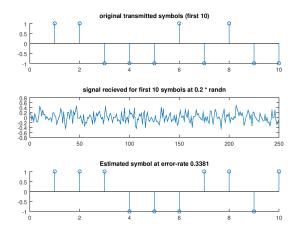


Figure 1.3: symbol detection using matched filter reciever

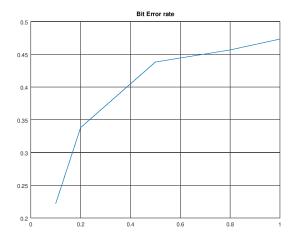


Figure 1.4: bit error rated plotted against variance parameteres

Aliasing considerations

2.1 aliasing in pulse

pulse wave form $p(t) = e^{-\pi t^2}$

$$\to P(f) = e^{-\pi f^2}$$

after sampling with a frequency f_s , the Fourier transform of the sampled sequence is

$$X_s(f) = \sum_k X(f - kf_s)$$

given that at zero frequency, aliasing should not be more than 10^{-2} i.e

$$\sum_{k \neq 0} X(f - kf_s) < 10^{-2}$$

$$\sum_{k \neq 0} e^{-\pi k^2 f s^2} < 10^{-2}$$

for the minimum value of f_s ,

$$2 * \sum_{k=1}^{\infty} e^{-\pi k^2 f s^2} < 10^{-2}$$

```
close all;
fs = 0:0.01:1;
k = 1:10;
y = zeros(1,length(fs));
temp = ones(1,10);
for i = 1:length(fs)
    for ki = k
        temp = exp(-1*pi*(ki*fs(i))^2);
    end
    y(1,i) = sum(temp);
end
plot(fs,y);
hold on;
tol = ones(1,length(fs))*0.01;
plot(fs,tol);
```

2.2. CODE matched filter

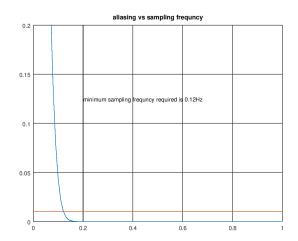


Figure 2.1: minimum fs to avoid aliasing

```
title("aliasing vs sampling frequncy");
ylim([0 0.2])
grid on;
idx = lookup(y,0.01);
text(0.2,0.125,["minimum sampling frequncy required is "num2str(fs(←
idx)) "Hz"]);
print -dpng "aliasing.png"
```

2.2 code

2.2. CODE matched filter

```
p_{trunc} = p((2*fs)-idx:(2*fs)+idx)/fs; %multiplication by Ts as \leftarrow
            dictated by impulse invariance method
      disp(["length" of the sequence P_trunc = "num2str(length(p_trunc)) \leftarrow
%modelling the low pass filter , taking the sampling freq(fs = 10) and←
      tolerance = 0.001
      {\tt tolerance} \, = \, 0.001
     RC = 0.5;
     t = 0:1/fs:3;
     h = \exp(-t/RC);
     idx = lookup(h, tolerance);
     h_{trunc} = 2*h(1:idx)/fs; %h is mutliplied by 2 to normalise the \leftarrow
      \begin{array}{lll} disp\left(["length of the sequence h at "num2str(fs)" and tolerance \leftrightarrow "num2str(tolerance)" is "num2str(length(h_trunc))]\right) \end{array}
      figure
           plot(t,h);
title("low pass filter at sending end");
%calculating the value of 'N' , given Rb = 0.4
     N = fs/0.4;
%generating random symbols for bpsk modulation
     k = 10000; %be the number of arbitrary symbols
     sk = -1 + 2*(rand(1,k)>0.5);
%generating pulses to be transmitted .
     mod_sig = [];
          for i = sk
                 \texttt{mod\_sig} \ = [\texttt{mod\_sig} \ i.*[\,\texttt{p\_trunc} \ \ \texttt{zeros}\,(1\,,\texttt{N-length}\,(\,\texttt{p\_trunc}\,)\,) \, \hookleftarrow
      endfor
Mplotting the modulated signal for first five symbol durations
           figure;
            subplot (2,1,1)
                 plot(mod_sig(1:5*N));
                  title(" first five symbols ");
%low pass filtering by h_n
      mod_sig_filtered = conv(h_trunc, mod_sig);
            subplot (2,1,2)
                 \textcolor{red}{\textbf{plot}} \left( \hspace{.08cm} \texttt{mod\_sig\_filtered} \left( \hspace{.08cm} 1 \hspace{.05cm} \colon \hspace{.08cm} 5 \hspace{.08cm} * \hspace{.08cm} \mathbb{N} \hspace{.08cm} \right) \hspace{.08cm} \right);
                  title ("modulated signal after low pass filtering")
                 print -djpg "modulatedsignal.jpg"
%addition of
                   gaussian noise
     len_tx_sig = length(mod_sig_filtered);
      tx_sig = \overline{zeros}(5, len_tx_sig);
     \begin{array}{l} \texttt{len\_tx\_sig} = \underbrace{\texttt{length} \left( \texttt{mod\_sig\_filtered} \right);} \\ \texttt{variance\_param} = \begin{bmatrix} 0.1 & 0.2 & 0.5 & 0.8 & 1.0 \end{bmatrix}; \\ \end{array}
     %multiplication by this parameter changes amplitude of the \leftrightarrow
            gaussian noise
      for i = 1:5
           \texttt{tx\_sig}\,(\texttt{i}\,,:) \; = \; \texttt{mod\_sig\_filtered} + \; \texttt{variance\_param}\,(\texttt{i}\,) \,.*\, \texttt{randn}\,(\texttt{1}\,, \hookleftarrow)
                  len_tx_sig);
%matched filter reciever
     {\tt basis} \, = \, {\tt conv} \, (\, {\tt p\_trunc} \, \, , \, {\tt h\_trunc} \, ) \, ;
      matched_filter = basis(end:-1:1);
%decoding of symbols
      sk_hat = zeros(5,k);
      for i = 1:5
           matched_filter_out = conv(tx_sig(i,:), matched_filter);
            sk_hat(i,:) = -1 + 2.*(matched_filter_out(2*N:length(p_trunc): \leftarrow
                  end-length(matched_filter))>0);
%calculation of bit error rate
      error_rate = zeros(1,5);
```

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