ECOLE POLYTECHNIQUE FEDERALE DE LAUSANNE

School of Computer and Communication Sciences

Software-Defined Radio: A Hands-On Course Midterm October 30, 2013

Problem 1. (8 p.)

```
% define the parameters
f1=100; f2=300;
Fs=1000;
NFFT=1024;
T=10;  %seconds

t=0:1/Fs:T;  % time vector
vf=-Fs/2:Fs/NFFT:Fs/2-Fs/NFFT;  %frequency vector

s=sin(2*pi*f1*t)+sin(2*pi*f2*t);

%plot the frequency spectrum
figure, plot(vf,abs(fftshift(fft(s,NFFT)))), xlabel('f[Hz]'), ylabel('|s_F(f)|')
```

Problem 2. (8 p.)

- (a) The signal s(t) is obtained by multiplying the sum of sinusoids by a rectangular window. The Fourier transform of a sinusoid is given by two symmetric Dirac functions. The Fourier transform of a rectangular pulse is a sinc. So, the Fourier transform of s(t) will be the convolution between four Dirac functions and a sinc.
- (b) The spectrum is represented from $-f_s/2$ to $f_s/2$, so $f_s = 1000$ Hz. The main lobes have width 2W = 40Hz and the secondary lobes have width W = 20Hz. From W = 1/T, we obtain T = 50ms.
- (c) The lobes are narrower for the second figure. As their width is inverse proportional to T, we conclude that T has increased.
- (d) Because the frequencies f_1 and f_2 did not change, but the location of the peaks changed, we conclude that the sampling frequency changed. The new value is $f_{s2} = 500$ Hz. In this case W = 10Hz, so T = 100ms.

Problem 3. (3 p.)

```
% xcorr;
ip1=xcorr(a,b);
ip1=ip1(length(a));

% conv
ip2=conv(a,fliplr(conj(b)));
ip2=ip2(length(a));

% vector multiplication
ip3=a*b';
```

The third method is the most efficient: there are no unnecessary computations. For xcorr and conv, we only need the middle value, and discard all the others.

Problem 4. (3 p.) Let **s** be the given sequence:

```
0 0 0 0 -1 0 -1 0 -2 2 2 -2 -3 0 1.
```

- (a) The zeroes are at the beginning of the sequence, so s1 is longer.
- (b) length(s1)+length(s2)=length(s) number of initial zeroes +1= 15-4+1=12;
- (c) Because s1 is the longest, length(s) will be 2*length(s1)-1. Therefore, length(s1)=8 and length(s2)=4.

```
Problem 5. (12 p.)
% SDR 2013 midterm - Problem 5
clear all;close all;clc;
%the number of transmitted symbols (not including the training sequence)
nSymbols=1000;
% load the training sequence, named traninig
load training;
%load the received signal containing the delay, the training symbols and
%the modulation symbols
load received
%% (a) estimate the start of the QAM symbol sequence in the received data and the
%rotation
corrResult=xcorr(received, training);
corrResult(1:length(received)-1)=[];
%start position of the training sequence
[M, pos] = max(abs(corrResult));
%start position of the QAM symbols
startPos=pos+length(training);
%estimate the rotation angle
rTraining=received(pos:pos+length(training)-1);
estimRotation=mean(rTraining./training);
%% (b) pull out the noisy symbol sequence and correct for the phase rotation
rSymbols=received(startPos:startPos+nSymbols-1);
rSymbolsCorrected=rSymbols*estimRotation';
%% (c) plot the complex plane noisy symbols
scatterplot(rSymbols), title ('before rotation correction')
scatterplot(rSymbolsCorrected), title ('after rotation correction')
%% (d) decode the bit sequence
bitsREstimated=sign(real(rSymbolsCorrected));
bitsIEstimated=sign(imag(rSymbolsCorrected));
% you can compare the result with the bit sequences stored in bitsR.mat
```

```
% and bitsI.mat
load bitsR
load bitsI
errR=nnz(bitsR-bitsREstimated)
errI=nnz(bitsI-bitsIEstimated)
Problem 6. (3 p.)
y_up = upsample(y, USF);
% the noise is complex
noise = sigma/sqrt(2) * (randn(size(z))+1j*randn(size(z)));
% the receiver matched filter is the time-inversed conjugate transmitter filter
y_rec = conv(r, fliplr(conj(h)));
Problem 7. (3 p.)
(a) %% determining the square of each element of vector X
    x2=x.^2;
(b) %% finding the position of the maximum element of vector X
    [^{\sim},pos]=max(x);
(c) %% mapping a vector of integers between 0 and 3, denoted X,
    % to QAM constellation points
    % you know that qammap=[-1+1i, -1-1i, 1+1i, 1-1i]
    x_qam = qammap(x+1);
```