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% EE 254
                                                                     Essam Marouf
% The LMS Algorithm: Example Adaptive Filters
                                                                     Matlab Handout #5
%
% Example 1. Simple Noise (Correlation) Canceler
% the primary (desired) signal d(k)=-0.8*x(k)+v(k), where both x(k)
% and v(k) are samples of white noise of zero mean and variance
\% = 0.1. The reference signal is xk.
%
clear, close all
LK=2000;
                                   % Sample indices
K = [1:LK];
                                   % noise standard deviation
sigma=sqrt(0.1);
shift =[0.2 \ 0.0 \ -0.2]; % plot shift
indx=0;
for mu=[0.01 0.03 0.1]
  indx=indx+1;
  x=sigma*randn(size(K)); % generates a vector of 200 samples of WGN of zero
                           % mean and sigma^2 variance
  v=sigma*randn(size(K)); % a new call to 'rand' generates a new random vector
  d = -0.8 * x + v;
                           % The average power of x is its variance =0.1;
                           % thus 0 < mu < 10; here mu << the upper bound so as
                           % to achieve reasonable statistical fluctuations.
  w0(1)=0;
                           % Filter has only one weight;
                           % initialized to an arbitrary value = 0.
%
% Iterate
%
  for k=1:1:LK
        y(k)=w0(k)*x(k);
                                                % Compute the filter output
        e(k)=d(k)-y(k);
                                                % Compare to the desired signal and
                                                % compute the error
        w0(k+1)=w0(k)+2*mu*e(k)*x(k);
                                                % Update the filter weight
  plot(K,w0(1:LK) + shift(indx)), grid, hold on
  w0=[];
title('A Simple Single Wight Noise (Correlation) Canceler'),
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xlabel('Iteration Number'), ylabel('Weight, w0'), hold off
```

- % Exponential convergence towards the theoretical limit $w^*=-0.8$
- % is evident. Also evident is the persistent random fluctuations
- % caused by the noisy gradient estimate. %
- % This correlation canceler removes the component of d(k) that is
- % correlated with x(k). In principle, this leaves the noise
- % component v(k) as the only component of the error signal e(k). To
- % check let's plot both v(k) and e(k) for k> 1000

%

% The following conclusions follow:

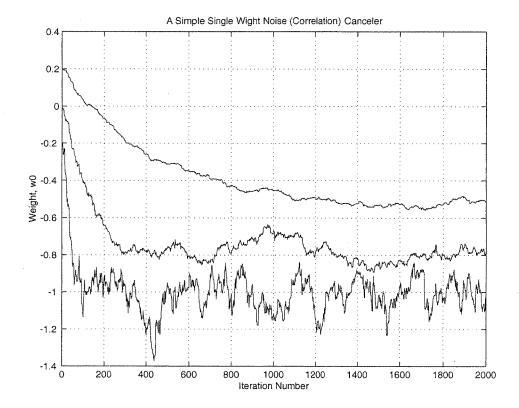
%

- % 1) The larger mu is, the faster the convergence is. The
- % theoretical wight time constants tau=1/(2*mu*lambda) for the three cases
- % mu = 0.1, 0.03, and 0.01 are 50, 167, and 500 samples, respectively.
- % This is evident in the figure.

%

- % 2) The larger mu is, the larger the statistical fluctuations in w0.
- % For example, the standard deviation of w0, computed for k > 1000
- % (StdZ above) is .0923,.0386 ,and .0283 , for mu=0.1, 0.03, and
- % 0.01, repectively. Thus a better statistical behavior is usually
- % obtained at the expense of slower adaptation.

%



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% Example 2. Interference Canceler; Deterministic Signals
<sup>0</sup>/<sub>0</sub> -----
%
clear, close all
              % sampling frequency in Hz
f_{S}=128:
f1=25;
              % interference frequency
f2=10;
              % frequency of interest
LK=1000;
x = 0.5*\cos(2*pi*(f1/fs)*[0:LK-1] + 18*pi/180);
d = cos(2*pi*(f2/fs)*[0:LK-1] + 63*pi/180);
d = d + \cos(2*pi*(f1/fs)*[0:LK-1] + 75*pi/180);
%
mu = 0.05;
                            % 7 weights are assumed (filter of order=6)
L=6:
                            % Initialize 7 weights to zero;
W=zeros(1,L+1)';
                             % We now handel W as a vector\
% Iterate
for k=L+1:1:LK
                            % X is also a vector containing x(k), x(k-1),
X=x(k:-1:k-L)';
                            % ..... X(K-L)
                            % As done in the theoretical formulation
y(k)=W'*X;
                            % Remember e is a scalar
e(k)=d(k)-y(k);
W=W+2*mu*e(k)*X;
                            % Update weight vector
end
% Plots of relevant results
% -----
%
NP1=1;
NP2=400;
subplot(4,1,1), plot([NP1:NP2],x(NP1:NP2)),
title('Reference Signal x(k)'),xlabel('k'),ylabel('x(k)'),
subplot(4,1,2), plot([NP1:NP2],d(NP1:NP2)),
title('Primary Signal d(k)'),xlabel('k'),ylabel('d(k)'),
subplot(4,1,3), plot([NP1:NP2],y(NP1:NP2)),
title('Filter Output Signal y(k)'),xlabel('k'),ylabel('y(k)'),
subplot(4,1,4), plot([NP1:NP2],e(NP1:NP2)),
title('Error Signal e(k)'),xlabel('k'),ylabel('e(k)'),
subplot,
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```
% Note that the filter prouces a copy of the the interference
% signal of frequency f1 = 25 \text{ Hz}. When subtracted from d, the
% output erros signal is a sinusoid of frequency f = 10 Hz, free
% of the interference.
%
b=W';
a=1;
[H,omega]=freqz(b,a,256);
%
tmp1=[f1/fs f1/fs],
tmp3=[0 \ 3];
tmp4=[-200 200];
figure
subplot(2,1,1),
plot(omega/(2*pi),abs(H),tmp1,tmp3,'--'),
grid, title('Frequency Response'),xlabel('f / fs'),ylabel('|H|'),
subplot(2,1,2),
plot(omega/(2*pi),angle(H)*180/pi,tmp1,tmp4,'--'),
grid, xlabel('f / fs'),ylabel('angle(H), deg'),
subplot
```

