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DSP Project 8
% Design of Two Channel Filter Banks
clear all; % clear workspace
clc;
            % clear command window
N = 49;
ws = 0.55*pi;
pointw = 400;
alpha = 1;
beta = 0.5;
epsilon = 0.00001;
9
wp = pi - ws;
NH = (N+1)/2;
deltaw = pi / pointw;
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P0 = zeros(NH, 1);
Qp = zeros(NH, NH);
Qs = zeros(NH, NH);
pointp = 0;
points = 0;
for iw = 0:pointw
   w = iw * deltaw;
   if w \le wp
      pointp = pointp + 1;
      C = zeros(NH, 1);
       for in = 1:NH
          C(in) = cos(w*(in - 0.5));
       end
       P0 = P0 - 2*C;
       Qp = Qp + C*C';
   elseif w >= ws
      points = points + 1;
       C = zeros(NH, 1);
       for in = 1:NH
          C(in) = cos(w*(in - 0.5));
       end
       Qs = Qs + C*C';
   end
end
P0 = wp * P0 / pointp;
Qp = wp * Qp / pointp;
Qs = (pi - ws) * Qs / points;
A = -0.5 * inv(Qp + Qs) * P0;
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h = zeros(N+1, 1);
h(1:NH) = 0.5 * A(NH:-1:1);
h(NH+1:N+1) = 0.5 * A;
FR = abs(freqz(h, 1, 0:deltaw:pi));
plot(0:deltaw/pi:pi/pi, FR);
xlabel('Normalized Frequency (\omega/\pi');
ylabel('Magnitude Response');
title('Initial Lowpass Filter');
90
deltak = 10000;
iteration = 0;
while deltak >= epsilon
   iteration = iteration + 1
   Ak1 = A:
   P = zeros(NH, 1);
   Q1 = zeros(NH, NH);
   Q2 = zeros(NH, NH);
   for iw = 0:pointw
     w = iw * deltaw;
     C = zeros(NH, 1);
      for in = 1:NH
         C(in) = cos(w*(in - 0.5));
      end
     Cpi = zeros(NH, 1);
      for in = 1:NH
         Cpi(in) = cos((w - pi)*(in - 0.5));
      P = P - 2*((Ak1'*C)*C + (Ak1'*Cpi)*Cpi);
      Q1 = Q1 + ((Ak1'*C)*C + (Ak1'*Cpi)*Cpi) * ((Ak1'*C)*C +
(Ak1'*Cpi)*Cpi)';
      if w >= ws
        Q2 = Q2 + C*C';
      end
   end
   P = pi * P / pointw;
   Q1 = pi * Q1 / pointw;
   Q2 = alpha * (pi - ws) * Q2 / points;
   A = -0.5 * inv(Q1 + Q2)*P;
   A = beta * A + (1 - beta) *Ak1;
   h = zeros(N+1, 1);
   h(1:NH) = 0.5 * A(NH:-1:1);
   h(NH+1:N+1) = 0.5 * A;
   FR = abs(freqz(h, 1, 0:deltaw:pi));
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subplot(1, 2, 1);
   plot(0:deltaw/pi:pi/pi, FR);
   axis([0, 1, 0, 1.1]);
   xlabel('Normalized Frequency (\omega/\pi');
   ylabel('Magnitude Response');
   title('Lowpass Filter');
   % frequency response of the whole system
   h1 = zeros(N+1, 1);
   for in = 0:N
      h1(in + 1) = (-1)^in * h(in + 1);
   end
   t = conv(h, h) - conv(h1, h1);
   FRt = abs(freqz(t, 1, 0:deltaw:pi));
   subplot(1, 2, 2);
   plot(0:deltaw/pi:pi/pi, FRt);
   axis([0, 1, 0.8, 1.2]);
   xlabel('Normalized Frequency (\omega/\pi');
   ylabel('Magnitude Response');
   title('Overall System');
   % stop condition
   deltak = norm(A - Ak1) / norm(A)
   pause;
end
% Two Channel Filter Banks
% deltak = 0.0750 (iteration = 1)
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