FIR FILTERS

$$H(z) = \sum_{n=0}^{N} h(n)z^{-n} = \frac{h(0)z^{N} + h(1)z^{N-1} + \dots + h(N-1)z + h(N)}{z^{N}}$$

LINEAR-PHASE FIR FILTERS

n = N/2, i.e., Linear-phase FIR filters exhibits either symmetry or antisymmetry around

$$h(n) = h(N-n), \quad n = 0, 1, ..., N,$$

and for the antisymmetic case,

$$h(n) = -h(N-n), \quad n = 0, 1, ..., N$$

Type I:
$$h(n) = h(N-n)$$
, N even
Type II: $h(n) = h(N-n)$, N odd
Type III: $h(n) = -h(N-n)$, N even
Type IV: $h(n) = -h(N-n)$, N odd

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shown in the figure for a linear-phase lowpass filthe passband and stopband, respectively. Typically, an FIR specification is expressed in terms of the zero-phase function $H_R(\omega T)$ as ter. The acceptable deviations are $\pm \delta_c$ and $\pm \delta_s$ in

Specification of Linear-Phase FIR Filters



Example

Synthesize a linear-phase FIR filter with $\omega_c T = 0.3\pi$, $\omega_s T = 0.6\pi$, $\delta_c =$ rithm remez.m 0.02, and $\delta_s = 0.0025$ (lowpass). Use McClellan-Parks-Rabiner's algo-

```
gd = grpdelay(h, 1, w*pi/180); % Group delay
                                               Attenu = -Mag;
                                                                        Mag = 20*log10(abs(H));
                                                                                                                                                                                                                                                                                                                                            deltas = [0.02 0.0025];
                                                                                                                                                                                                                                                                                                                                                                                                      wT = [0.3 0.6]*pi;
                  Phase = angle(H)*180/pi;
                                                                                                     H = freqz(h, 1, w*pi/180);
                                                                                                                                  w = linspace(0, 180, 180*points+1);% wT axis
                                                                                                                                                                    points = 16;
                                                                                                                                                                                            h = remez(N, Be, D, W);
                                                                                                                                                                                                                                                                                                                  fsample = 2*pi;
                                                                                                                                                                                                                                                                                                                                                                           bands = [1 0];
                                                                                                                                                                                                                                                                                                                                                                                                                                     % Example Linear-Phase, lowpass FIR
                                                                                                                                                                                                                                                                                    [N, Be, D, W] = remezord(wT, bands, deltas, fsample);
                  % Phase response [deg]
                                                                             % Magnitude response [dB]
                                                                                                          % Transfer function
                                                                                                                                                                                                                                                          % Estimated filter order
                                                                                                                                                                                                                                                                                                                                                                              % Gain in the bands
                                                                                                                                                                                                                                                                                                                                                                                                         % Band edges
                                                % Attenuation [dB]
                                                                                                                                                                                                                                                                                                                                                % Acceptable deviations
```



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ω

In this case we get from remezord

N = 13 $Be = [0 \ 0.3 \ 0.6 \ 1]^{\mathrm{T}}$ $D = [1 \ 1 \ 0 \ 0]^{\mathrm{T}}$ $W = [1 \ 8]^{T}$

a filter that satisfies the specification. The filter order *N* that is obtained from **remezord** is only an estimation It may therefore be necessary to try different filter orders in order to find

 $N = 13 \Longrightarrow A_{min} \approx 51.7 \text{ dB and } A_{max} \approx 0.366 \text{ dB}$ $N = 14 \Longrightarrow A_{min} \approx 50.9 \text{ dB and } A_{max} \approx 0.384 \text{ dB}$ $N = 15 \Longrightarrow \text{neither acceptable}$ $N = 16 \Longrightarrow A_{min} \approx 54.9 \text{ dB and } A_{max} \approx 0.26 \text{ dB}.$

The symmetric impulse response is

h(0) = -0.003428964 = h(16) h(1) = 0.002941935 = h(15) h(2) = 0.021664496 = h(14) h(3) = 0.015363454 = h(13) h(4) = -0.041601459 = h(12) h(5) = -0.067766346 = h(11) h(6) = 0.061181730 = h(10)h(7) = 0.303702325 = h(9)

h(8) = 0.430966079



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Half-Band FIR Filters

filter is antisymmetric with respect to $\pi/2$, i.e., If the zero-phase function of an **even**-order lowpass, linear-phase FIR

$$H_R(e^{j\omega T}) = 1 - H_R(e^{j(\pi - \omega T)})$$

one in the center, which is 0.5. then every other coefficient in the impulse response is zero except for the

Delay-Complementary FIR Filters

A pair of filters, H(z) and $H_c(z)$ are delay-complementary if

$$H(z) + H_c(z) = z^{-M}$$

where *M* is a nonnegative integer.



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Transposed direct form FIR structure



Direct form FIR structure

 $y(n) = \sum_{k=0}^{N} h(k)x(n-k)$

FIR Structures

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Direct form linear-phase FIR structure 2M = N+1

 $y(n) = \sum_{k=0}^{N} h(k) [x(n-k) \pm x(n-N+k)]$



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x(n)

h(0)

Delay-Complementary FIR structure N = even and 2M = N-1

