

## Lecture 6 FIR Filter Implementations

### Objectives:

- Understand basics of FIR filter design using MATLAB
- Understand C67x implementations
- Understand circular buffer usage

## FIR Filter Basics

- The transfer function of a  $M^{\text{th}}$  order FIR filter is:

$$H(z) = b_0 + b_1 z^{-1} + \dots + b_M z^{-M}$$

- The difference equation is:

$$y(n) = b_0 x(n) + b_1 x(n-1) + \dots + b_M x(n-M)$$

- For linear phase FIR filters, the coefficients are real & symmetrical
- FIR filters are easy to implement in either hardware or DSP software
- FIR filters are inherently stable

## FIR Filter Design in MATLAB

- MATLAB provides a large variety of routines for designing FIR filters. We will concentrate on Parks-McClellan's algorithm.

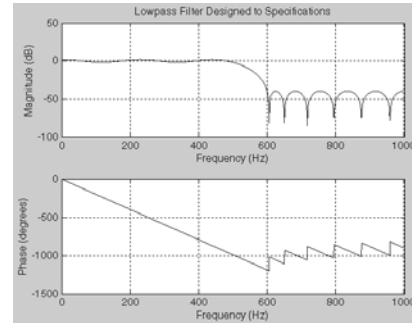
FIR Filter Design	Description
<code>cremez</code>	Complex and nonlinear-phase equiripple FIR filter design
<code>fir1</code>	Design a window-based finite impulse response filter
<code>fir2</code>	Design a frequency sampling-based finite impulse response filter
<code>fircls</code>	Constrained least square FIR filter design for multiband filters
<code>fircls1</code>	Constrained least square filter design for lowpass and highpass linear phase FIR filters
<code>firls</code>	Least square linear-phase FIR filter design
<code>firrcos</code>	Raised cosine FIR filter design
<code>intfilt</code>	Interpolation FIR filter design
<code>kaiserord</code>	Estimate parameters for an FIR filter design with Kaiser window
<code>remez</code>	Compute the Parks-McClellan optimal FIR filter design
<code>remezord</code>	Parks-McClellan optimal FIR filter order estimation
<code>sgolay</code>	Savitzky-Golay filter design

## REMEZORD & REMEZ functions

- $[N, F_0, M_0, W] = \text{REMEZORD}(F, M, \text{DEV}, F_s)$ 
  - finds the approximate order  $N$ , normalized frequency band edges  $F_0$ , frequency band magnitudes  $M_0$  and weights  $W$  to be used by the REMEZ function as follows:
 
$$B = \text{REMEZ}(N, F_0, M_0, W)$$
- The resulting filter will approximately meet the specifications given by the input parameters  $F$ ,  $M$ , and  $\text{DEV}$ .
- $F$  is a vector of cut-off frequencies in Hz, in ascending order between 0 and half the sampling frequency  $F_s$ .
- $M$  is a vector specifying the desired function's amplitude on the bands defined by  $F$ .
- The length of  $F$  is twice the length of  $M$ , minus 2 (it must therefore be even).
- The first frequency band always starts at zero, and the last always ends at  $F_s/2$ .
- $\text{DEV}$  is a vector of maximum deviations or ripples allowable for each band.

## A Design Example

- Lowpass filter spec:
  - 500 Hz passband cutoff frequency, less than 3 dB passband ripple
  - 600 Hz stopband cutoff frequency, with at least 40 dB attenuation
  - sampling frequency of 2000 Hz



```

rp = 3;          % Passband ripple
rs = 40;        % Stopband ripple
fs = 2000;      % Sampling frequency
f = [500 600]; % Cutoff frequencies
a = [1 0];      % Desired amplitudes

% Compute deviations
dev = [(10^(rp/20)-1)/(10^(rp/20)+1) 10^(-rs/20)];

[n,fo,ao,w] = remezord(f,a,dev,fs);
b = remez(n,fo,ao,w);
freqz(b,1,1024,fs);
title('Lowpass Filter Designed to Specifications');
    
```

## FIR Implementation on C6711

- Must declare buffer arrays to store input samples  $x(n)$ , and coefficients  $b(n)$ , for  $n = 0$  to  $M$

```

#define M      12
// FIR filter coefficients
float b[] = {1.0, 0.2356, -0.03987,
            .....
            };
float x[M];
    
```

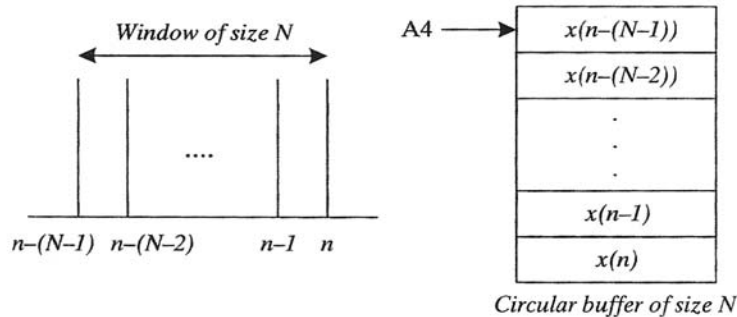
- Require to implement delay operator by shifting all previous  $M-1$  samples and insert current input sample in  $x[0]$ :

```

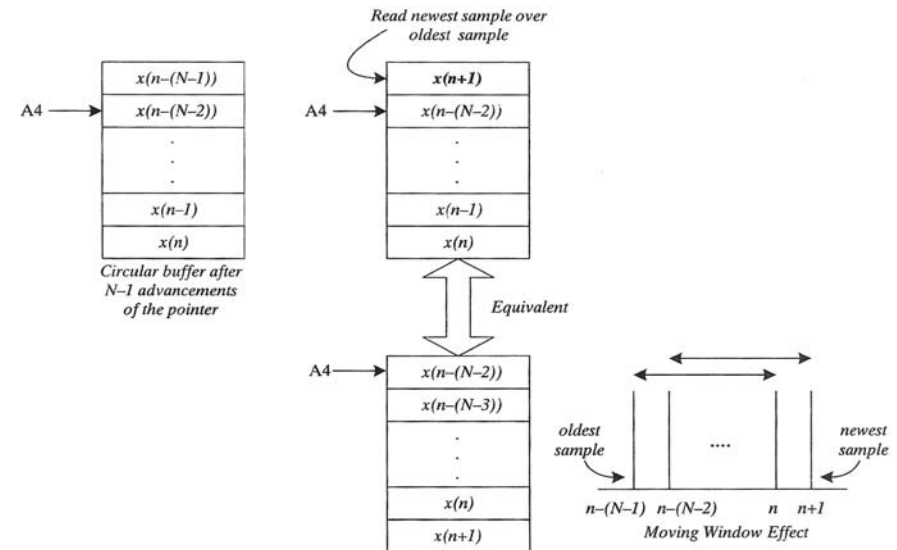
// Z-1 delay operator
for (i=M-2; i>=0; i--) x[i+1] = x[i];
x[0] = input_sample;
    
```

## Circular Buffers

- Shifting data this way is a form of circular buffer - inefficient
- Far better to implement this circular buffer using fixed memory (without moving data) and a moving pointer
- After  $N$  accesses,  $A4$  returns to the same place

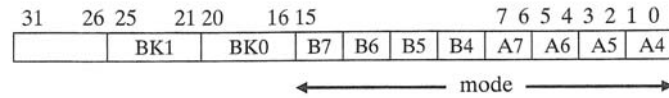


## Circular buffer with sliding window



## C6x Support for circular buffers

- Program Address Mode Register AMR according to:



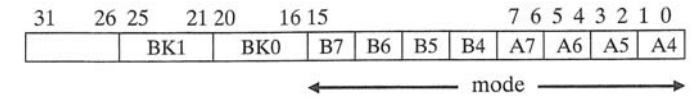
BK0/BK1 = N  
Block size  
(bytes) =  $2^{N+1}$

Mode  
00: linear (default)  
01: circular (using BK0)  
10: circular (using BK1)  
11: reserved

- BK0 and BK1 defines the block size of the buffer as  $2^{(N+1)}$  bytes
- C6x allows two independent circular buffer sizes in powers of 2
- Choose which 8 registers (A4-7, B4-7) as circular buffer pointers

## C6x Support for circular buffers setup

- Example: A4 as a pointer for buffer size of 256 bytes



BK0/BK1 = N  
Block size  
(bytes) =  $2^{N+1}$

Mode  
00: linear (default)  
01: circular (using BK0)  
10: circular (using BK1)  
11: reserved

```
asm ("MVK.S2 0001h,B2"); // A4 used as pointer
asm ("MVKLH.S2 0007h,B2"); // use BK0, size is 256
asm ("MVC.S2 B2,AMR"); // setup AMR
```