

# Discrete Time FIR Filtering

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# Introduction: FIR Filters in Digital Signal Processing

## What are FIR Filters?

- **Finite Impulse Response (FIR):** Filter output depends only on current and past inputs
- Impulse response has *finite* duration (eventually becomes zero)
- Fundamental building block in discrete-time signal processing

## General Form:

$$y[n] = \sum_{k=0}^M b_k x[n-k] = b_0 x[n] + b_1 x[n-1] + \cdots + b_M x[n-M]$$

where  $h[n] = b_n$  for  $0 \leq n \leq M$  (impulse response coefficients)

## Key Advantages:

- **Always stable** (finite sum cannot diverge)
- **Exact linear phase possible** (symmetric delay, no distortion)
- **Simple structure** (no feedback, easy to implement)

# Discrete-Time Implementation of FIR Filters

DSP

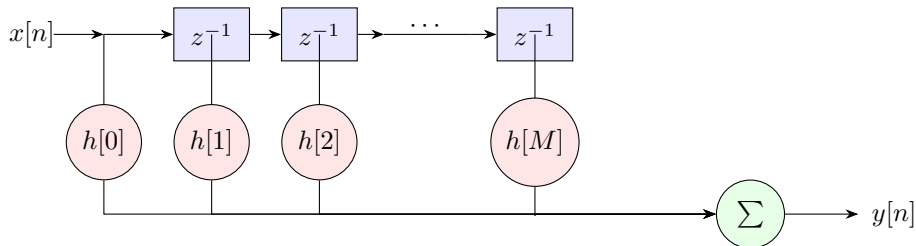
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**Difference Equation:**

$$y[n] = \sum_{k=0}^M h[k] x[n - k]$$

**Computational Cost (per output sample):**

- **Multiplications:**  $M + 1$
- **Additions:**  $M$
- **Memory:**  $M$  delay elements +  $(M + 1)$  coefficients

# Four Types of Linear-Phase FIR Filters

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## Classification by Symmetry and Length:

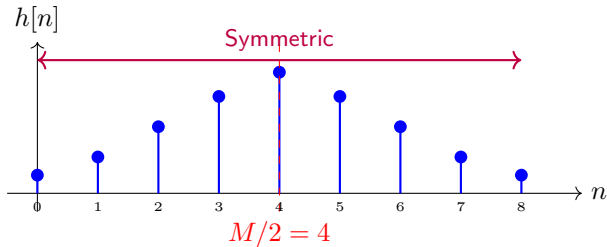
Type	Length	Symmetry	Applications
I	$(M + 1)$ odd $M$ even	Symmetric $h[n] = h[M - n]$	Lowpass, highpass, bandpass Most versatile
II	$(M + 1)$ even $M$ odd	Symmetric $h[n] = h[M - n]$	Lowpass, bandpass NOT for highpass
III	$(M + 1)$ odd $M$ even	Antisymmetric $h[n] = -h[M - n]$	Differentiators Hilbert transformers
IV	$(M + 1)$ even $M$ odd	Antisymmetric $h[n] = -h[M - n]$	Differentiators Hilbert transformers

## Key Distinction:

- **Symmetric** (Types I & II): Real-valued frequency response
- **Antisymmetric** (Types III & IV): Imaginary frequency response

# Type I: Symmetric, Odd Length

## Impulse Response Example ( $M = 8$ , length = 9):



## Frequency Response:

$$H(e^{j\omega}) = e^{-j\omega M/2} \underbrace{\left[ h[M/2] + 2 \sum_{n=1}^{M/2} h[M/2 - n] \cos(\omega n) \right]}_{\text{Real amplitude } A_e(e^{j\omega})}$$

## Type II: Symmetric, Even Length

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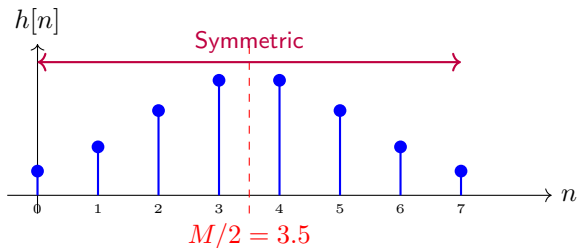
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**Impulse Response Example ( $M = 7$ , length = 8):**



**Frequency Response:**

$$H(e^{j\omega}) = e^{-j\omega M/2} \cos(\omega/2) \underbrace{P(\cos \omega)}_{\text{Polynomial}}$$

**Limitation:**  $\cos(\pi/2) = 0$  forces  $H(e^{j\pi}) = 0$  (cannot realize highpass)

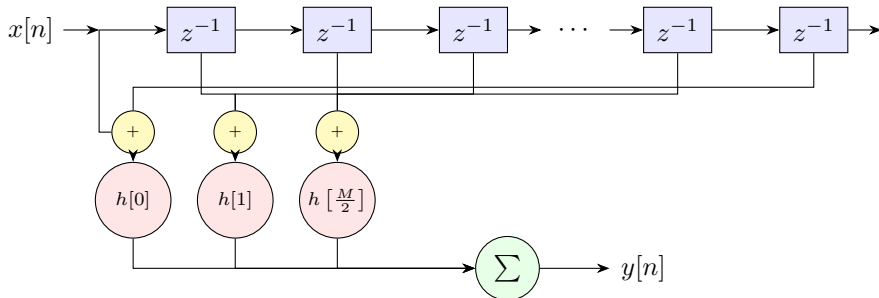
# Efficient Implementation: Exploiting Symmetry

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## For Symmetric FIR Filters (Types I & II):

Since  $h[n] = h[M - n]$ , we can reduce computations by 50%



## Optimized Cost: Trading Multiplies for Additions

- **Multiplies:**  $\lceil (M + 1)/2 \rceil$  (50% reduction!)
- **Additions:**  $M$  (pre-additions) +  $\lceil (M + 1)/2 \rceil - 1$  (post-additions)
- **Memory:** Still  $M$  delays +  $(M + 1)$  coefficients

## Type III: Antisymmetric, Odd Length

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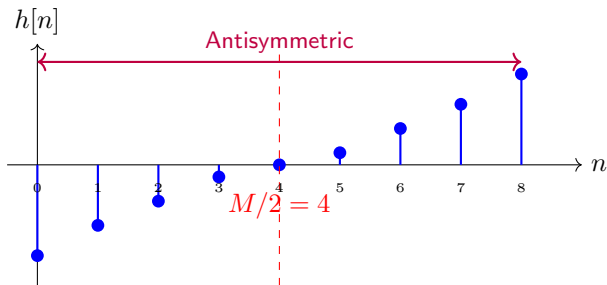
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**Impulse Response Example ( $M = 8$ , length = 9):**



**Frequency Response:**

$$H(e^{j\omega}) = je^{-j\omega M/2} \sin(\omega) P(\cos \omega)$$

**Limitation:**  $\sin(0) = 0$  forces  $H(e^{j\pi 0}) = 0$  (cannot realize lowpass)



## Type IV: Antisymmetric, Even Length

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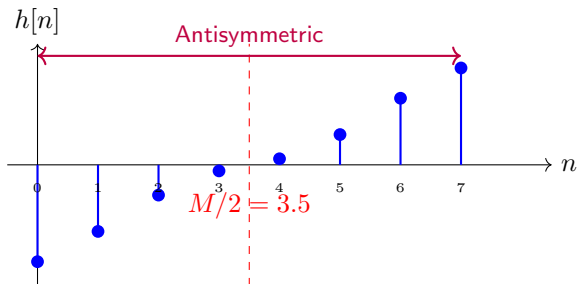
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**Impulse Response Example ( $M = 7$ , length = 8):**



**Frequency Response:**

$$H(e^{j\omega}) = je^{-j\omega M/2} \sin(\omega/2) P(\cos \omega)$$

**Limitation:**  $\sin(0) = 0$  forces  $H(e^{j\pi(0)}) = 0$  (cannot realize lowpass)

# Comparison: All Four Types

Property	Type I	Type II	Type III	Type IV
$M$	Even	Odd	Even	Odd
Length ( $M + 1$ )	Odd	Even	Odd	Even
Symmetry	Sym	Sym	Antisym	Antisym
$h[M/2]$	Non-zero	N/A	Zero	N/A
$H(0)$	Any	Any	Zero	Non-zero
$H(\pi)$	Any	Zero	Zero	Non-zero
Lowpass	✓	✓	×	×
Highpass	✓	×	×	✓
Bandpass	✓	✓	×	✓
Differentiator	×	×	✓	✓
Hilbert	×	×	✓	✓

## Design Choice:

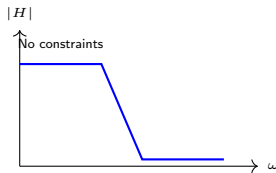
- Type I: Most versatile, use for general frequency-selective filters
- Type II: Lowpass only (response forced to zero at  $\omega = \pi$ )
- Types III & IV: Specialized applications (differentiators, Hilbert)

# Frequency Response Constraints

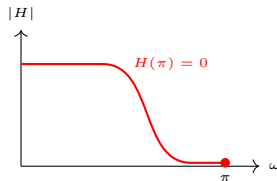
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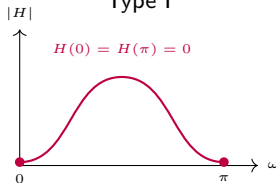
Why certain applications are prohibited:



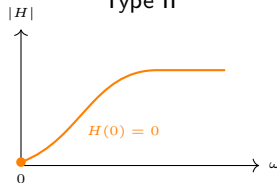
Type I



Type II



Type III



Type IV

# FIR vs IIR Filters: Implementation Comparison

Property	FIR Filters	IIR Filters
Stability	<b>Always stable</b> (no feedback)	<b>Can be unstable</b> (pole locations critical)
Phase Response	<b>Exact linear phase</b> possible (Types I-IV)	<b>Nonlinear phase</b> (except Bessel approximation)
Filter Order	<b>Higher order</b> needed (typically 10-100+ taps)	<b>Lower order</b> (typically 2-10 poles)
Computation	<b>More operations</b> per sample ( $M + 1$ multiplies)	<b>Fewer operations</b> (typically $< 10$ multiplies)
Memory	<b>More memory</b> ( $M$ delays)	<b>Less memory</b> (order of filter)
Finite Wordlength	<b>Robust</b> (no limit cycles, low sensitivity)	<b>Sensitive</b> (limit cycles, pole location errors)
Design	<b>Systematic</b> (windowing, Parks-McClellan)	More complex (bilinear transform, pole placement)

# When to Use FIR vs IIR Filters

## Choose FIR When:

- **Linear phase required**
  - Audio processing
  - Image processing
  - Data transmission
- **Stability critical**
  - Safety systems
  - Embedded systems
- **Fixed-point implementation**
  - Low wordlength DSPs
  - Quantization robustness needed

## General rule of thumb:

- Use **FIR** for audio/video (linear phase) and when stability/robustness are critical
- Use **IIR** for control systems and when computational resources are constrained

## Choose IIR When:

- **Computational efficiency critical**
  - Real-time constraints
  - Low-power devices
  - High sample rates
- **Sharp transitions needed**
  - Narrow stopband
  - Steep rolloff
  - Lower order achievable
- **Memory limited**
  - Small coefficient storage
  - Few delay elements