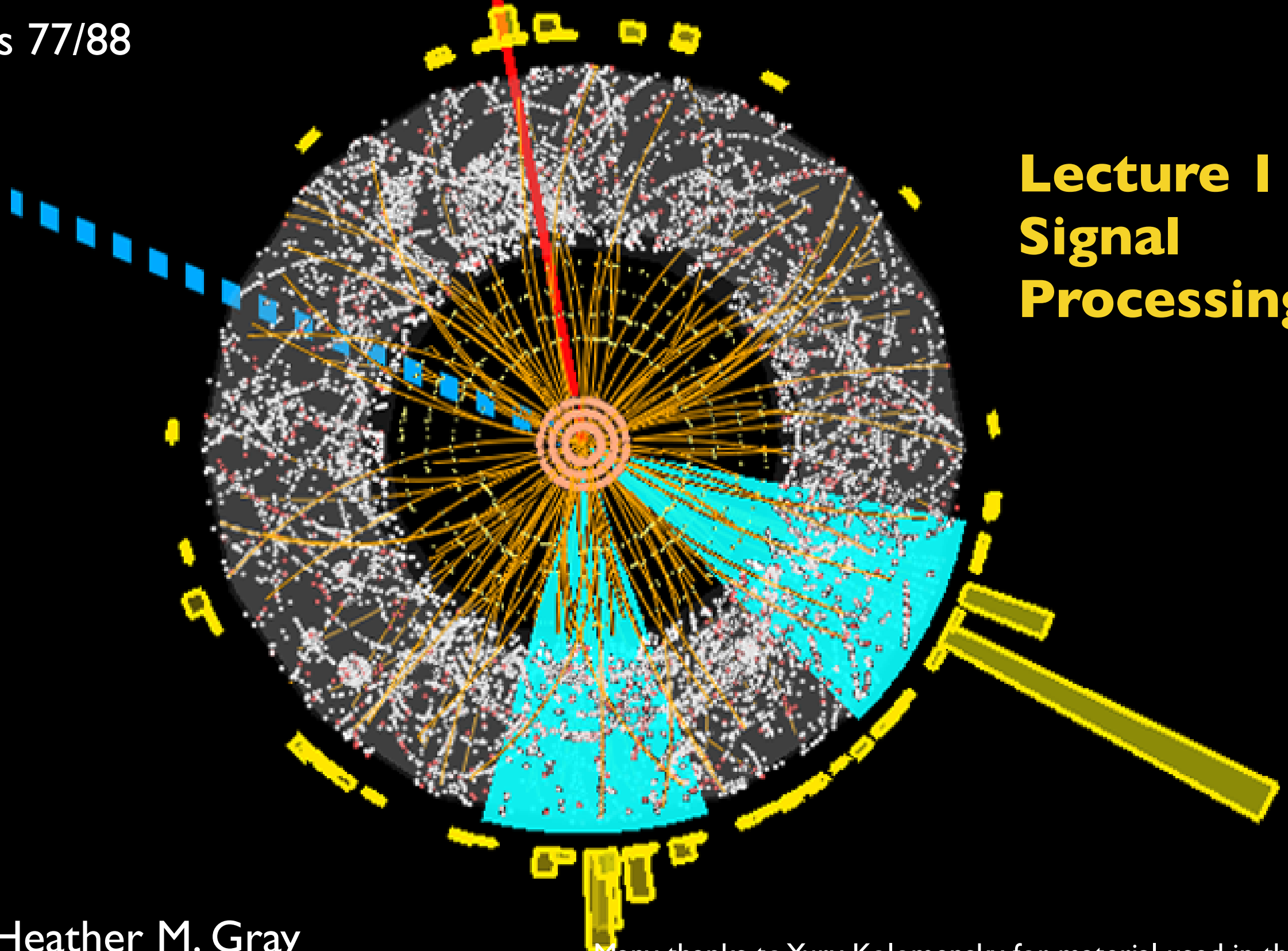


Introduction to Computational Techniques in Physics/Data Science Applications in Physics

Physics 77/88

Lecture 11: Signal Processing

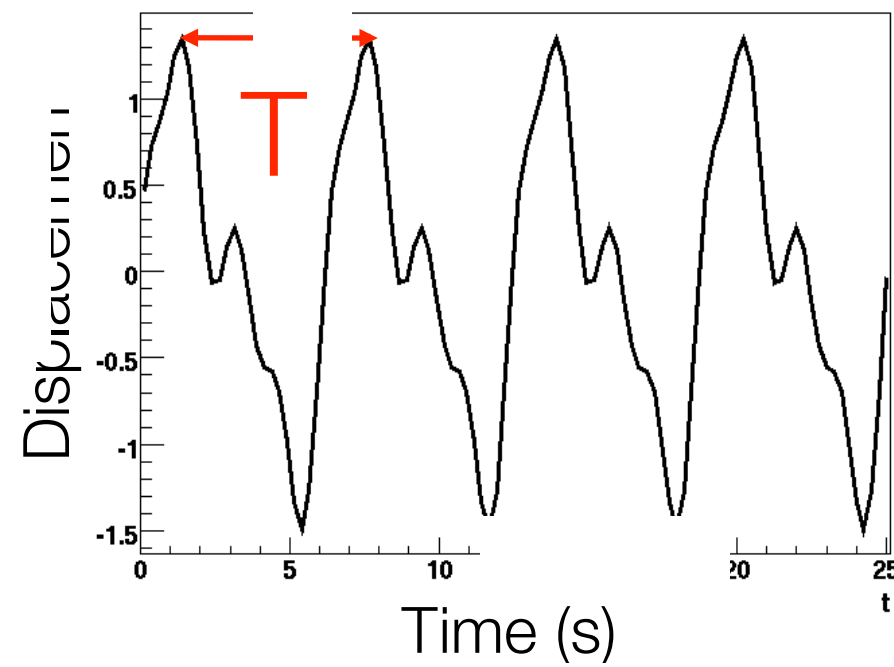


Prof. Heather M. Gray

Many thanks to Yury Kolomensky for material used in this course

Definitions

- Suppose $x(t)$ is some function that we sample at some frequency f_s
 - Measure $x(t)$ at times $t_n = nT$
 - Goal is to analyze $x(t)$ and f_s properties of $x(t)$
 - $T = 1/f_s$
 - f_s is the sampling frequency
 - T is the sampling period



Acronyms

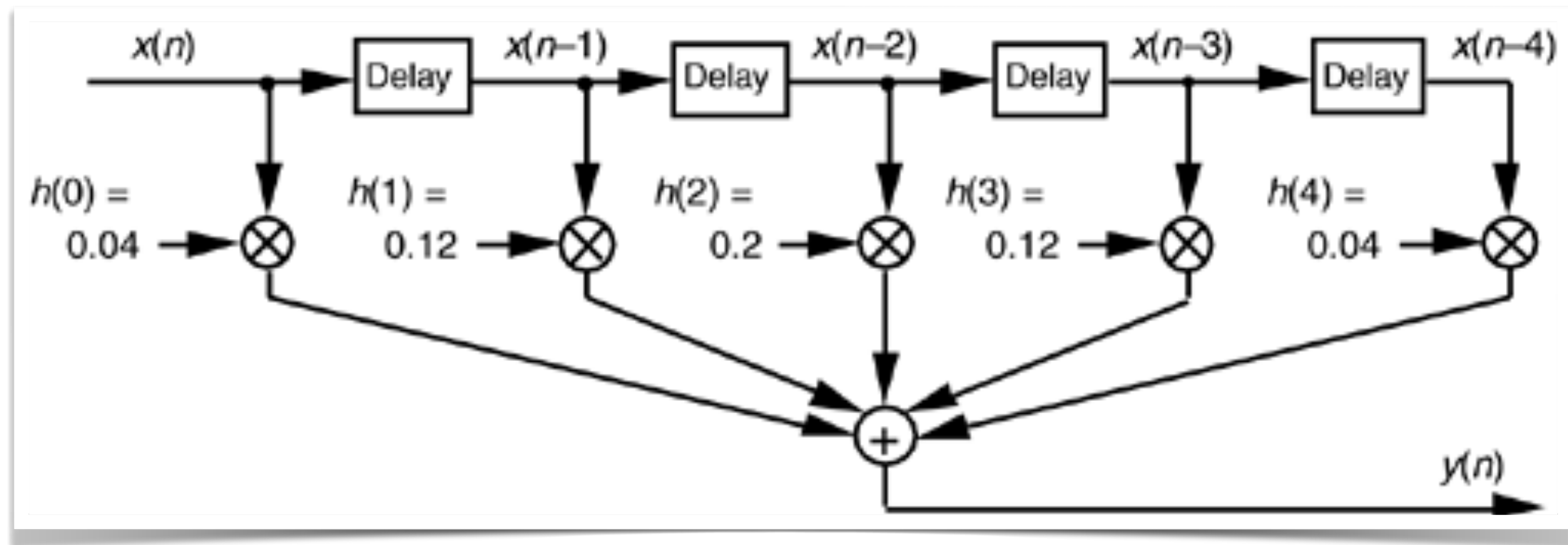
- FFT =
- DFT =
- DSP =
- SNR =
 - Often expressed in dB ,
 - i.e. $\text{SNR} = 10 \log_{10} \left(\frac{S}{N} \right)$ where S and N are amplitudes and
 - For power, $\text{SNR} = \frac{S}{N}$
- ASD =
 - PSD =
 - NPS =

Acronyms

- FIR =

- IIR =

Schematic Notation



- Represent operations

- Most can be this way

- i.e.

Fourier Transform

- Most common way to represent a continuous-time signal $x(t)$ is by means of a complex-valued function of frequency ω .

- Represent as a sum of sinusoids of different frequencies.

$$x(t) = \int_{-\infty}^{\infty} X(\omega) e^{j\omega t} d\omega$$

- Here $X(\omega)$ is a complex-valued function of frequency ω that represents the amplitude and phase of the sinusoids of the signal $x(t)$ at a given frequency ω .

- Also known as the Fourier transform.

Fourier Transform

- Inverse transform

- $x(\omega) = C \int$

- $x(f) = C \int$

- $=$

Discrete Fourier Transform

- For a $x(n)$, replace the n with a m

- $x(m) = \sum_{k=-\infty}^{\infty} x(k) \delta(m-k)$

- Define

- $x_{mag}(m) = |x(m)|$

- $\Delta\phi(m) = \angle x(m) - \angle x(m-1)$

- $P(m) = |x(m)|^2$

Aliasing

- For real signals, can show that
 - Exercise for the reader
 - $x(m) =$
 - Moreover (obvious)
 - $x(m) =$
- This is called
- Spectrum for is with
 - theorem:
 - can be reconstructed from
 - iff is limited to the and
 - is often called the frequency

Fast Fourier Transform

- Brute-force transforms can be
 - Require $O(N^2)$ calculations
- However, FFTs are commutative and obey the distributive property, which is used in the
 - Scale as $O(N \log N)$
 - Multiple algorithms exist, and are implemented in
 - E.g. Cooley-Tukey from 1965

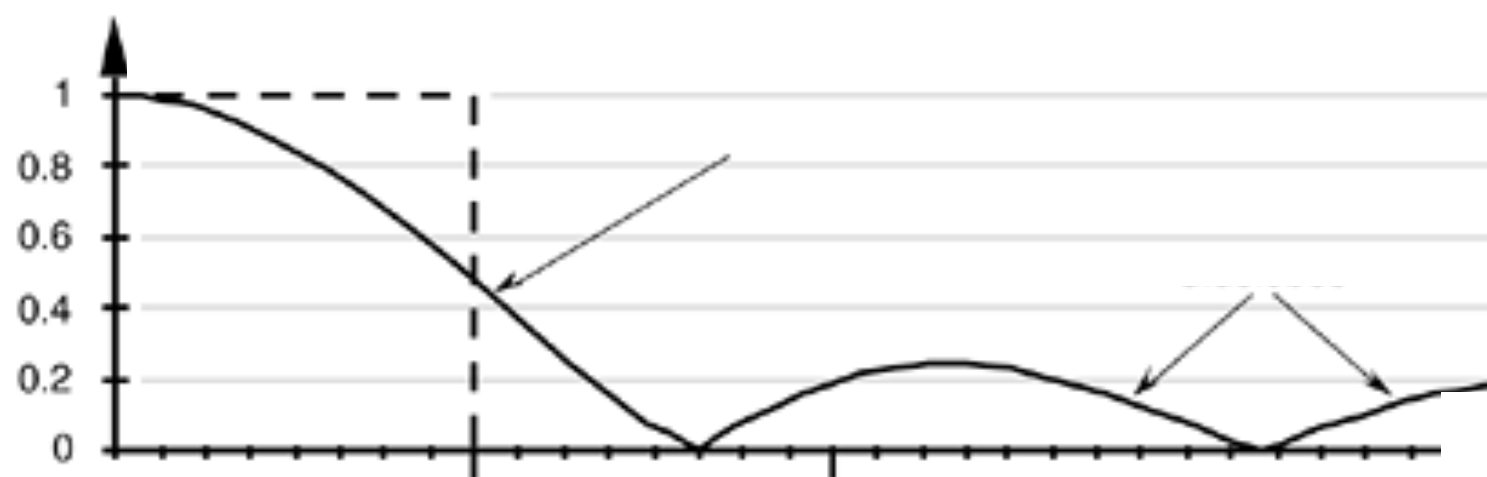
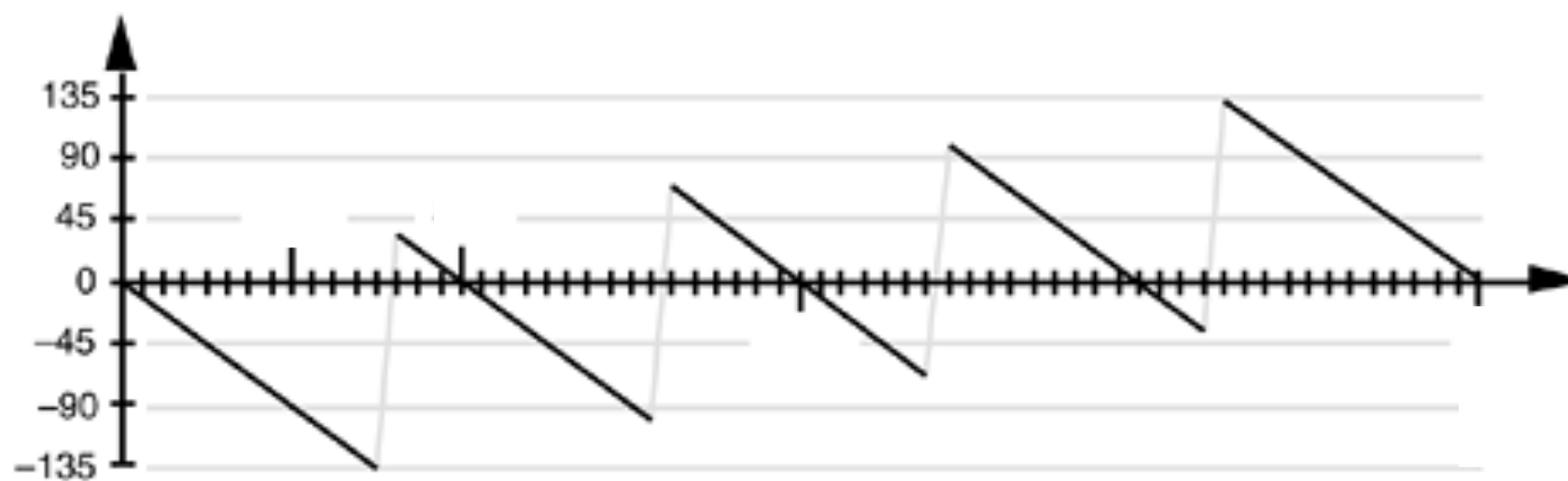
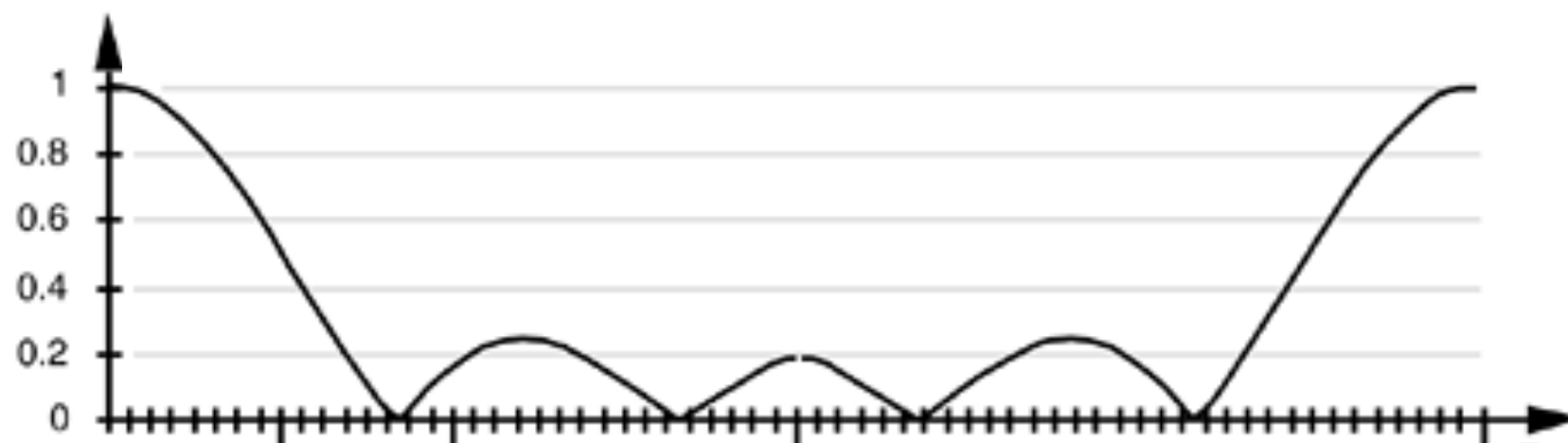
Applications of FFT

- FFT is useful to
 - Look at the $\text{amplitude spectrum}$ of a
 - e.g.
 - $\text{Power Spectral Density (PSD)}$
 - Spectral Density
 - Spectral Density
 - Look at the phase spectrum of
 - Phase Spectrum
 - Phase Spectrum
 - Phase Spectrum
 - Signal processing
 - Measure $\text{the power of the signal}$ in
 - the time domain
 - Design filters to
 - $\text{remove unwanted components}$

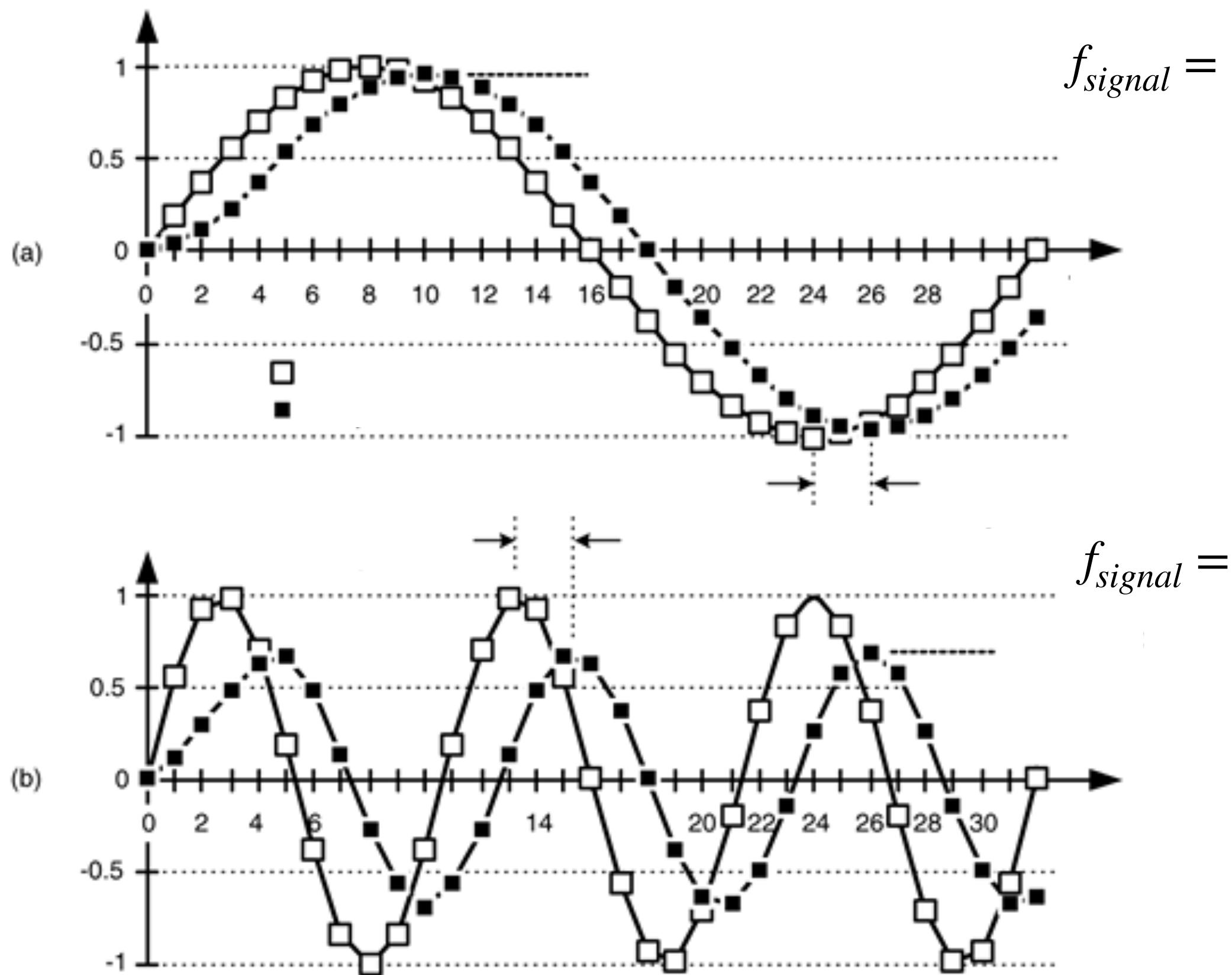
FIR filters

- is a way to the contributions of (random) to the of the
- Usually by with some (predetermined)
- Equivalent to

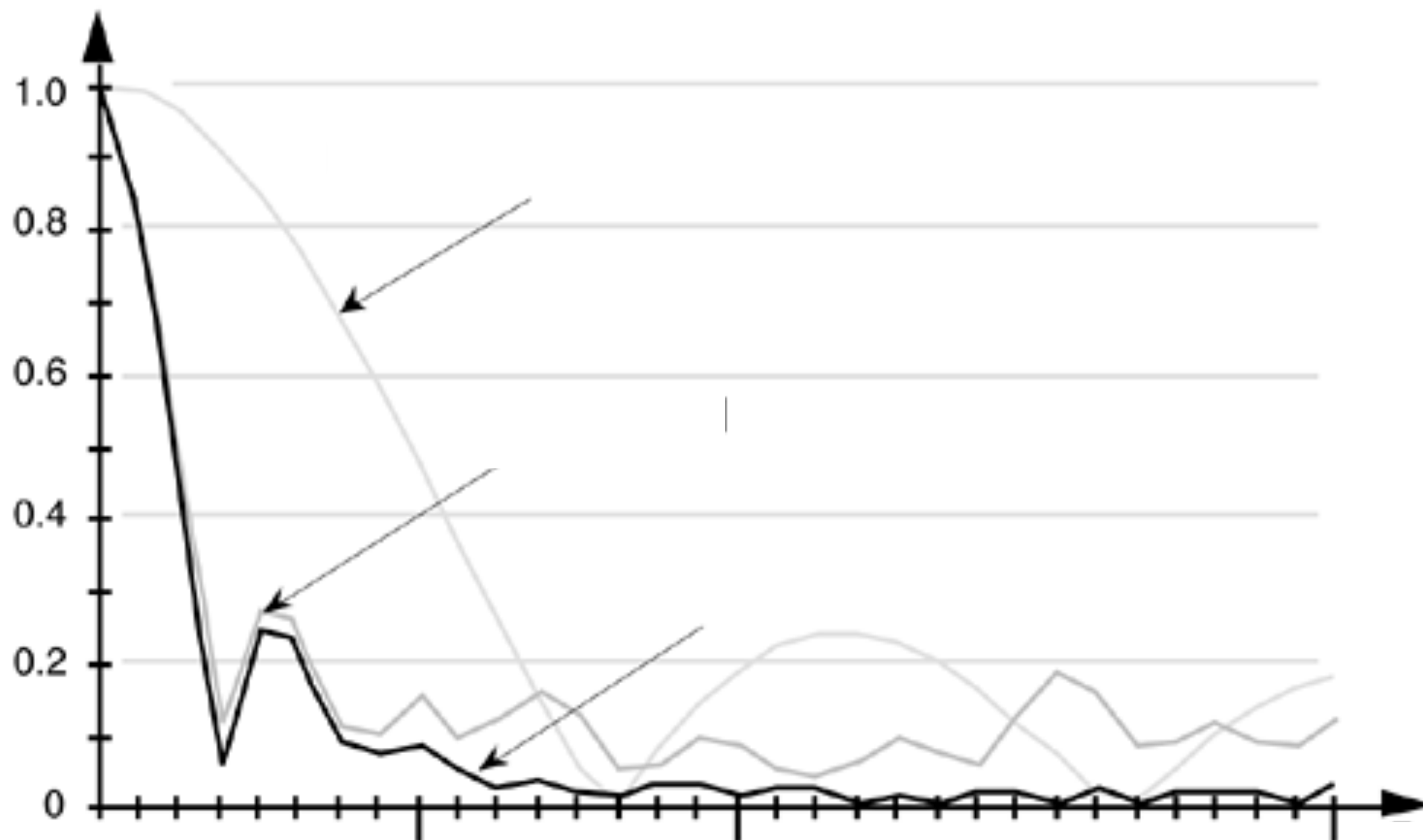
Box-car Filter Response



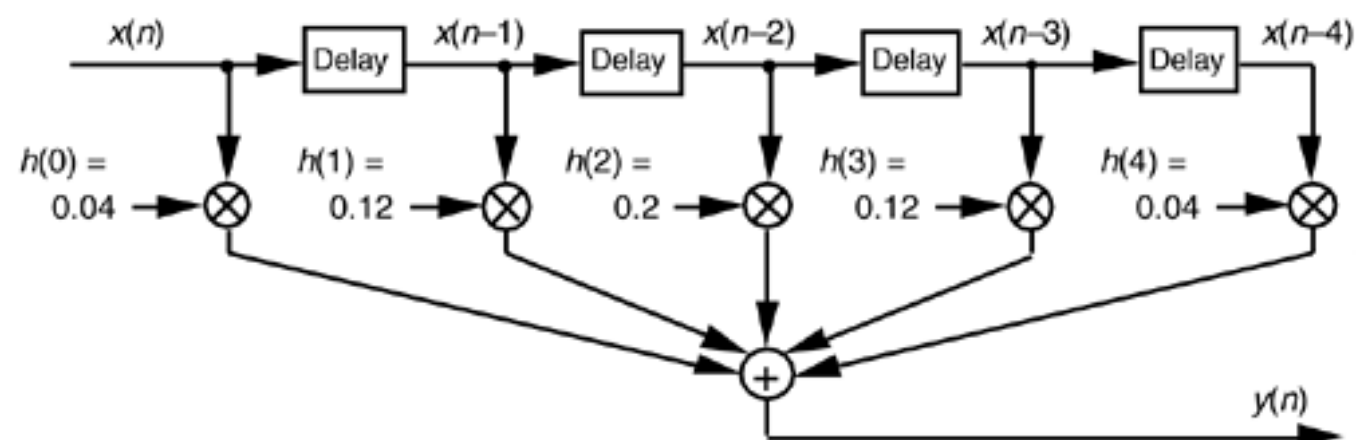
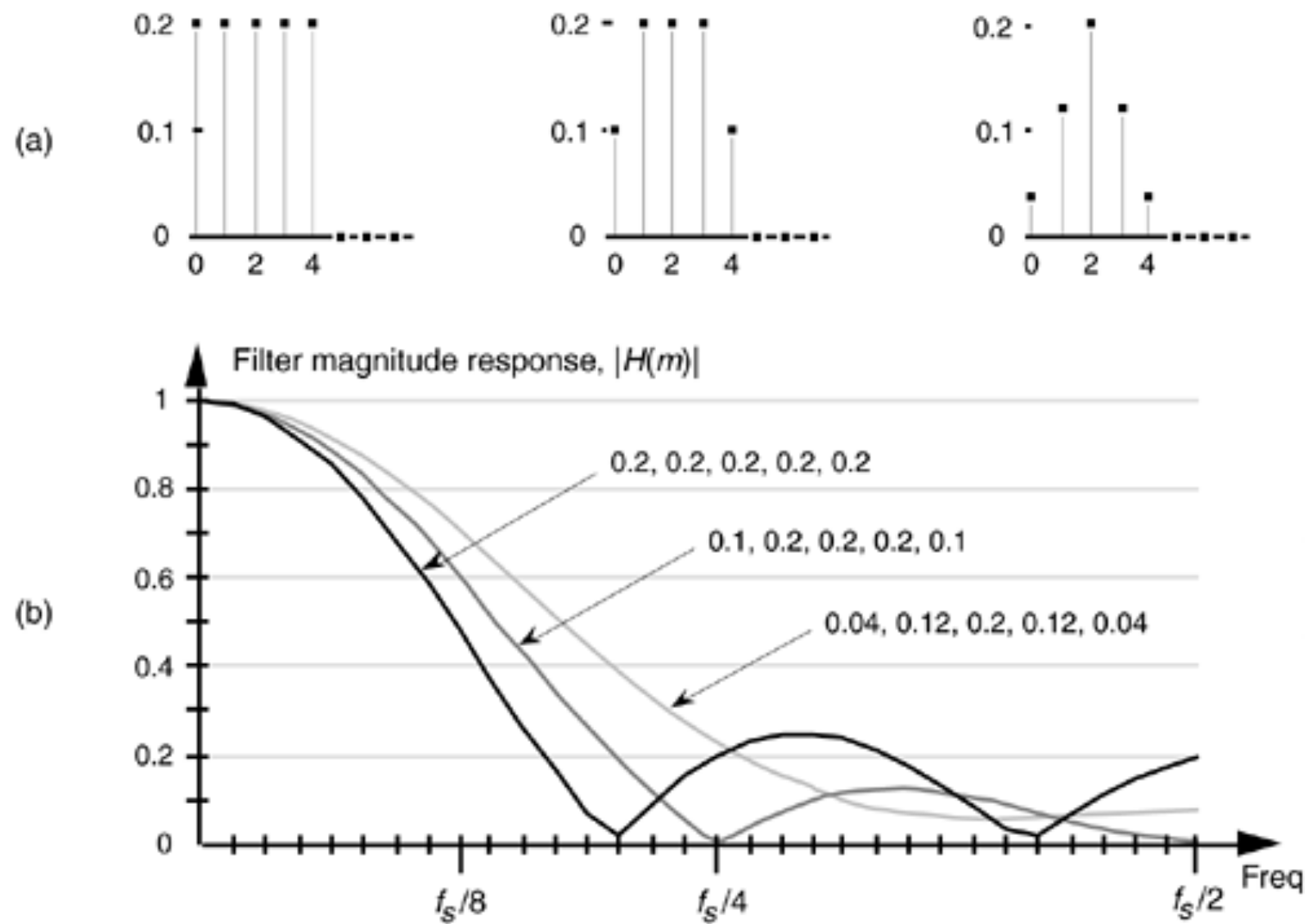
Time-Domain Signal After Filtering



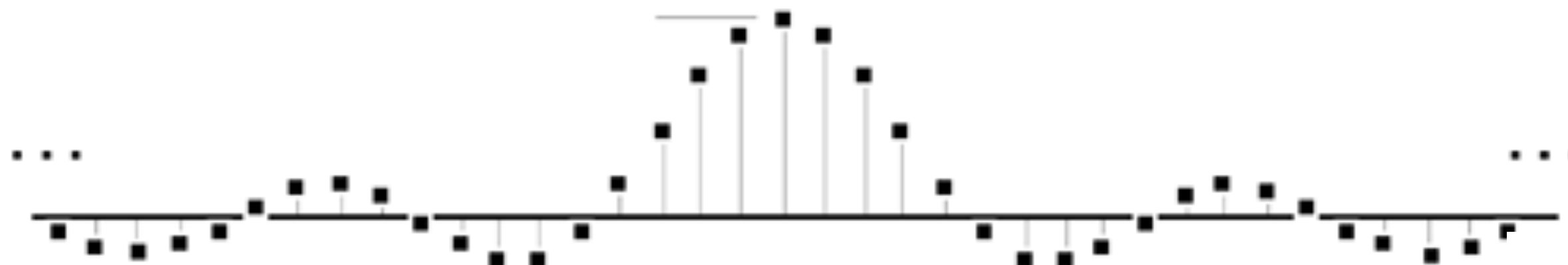
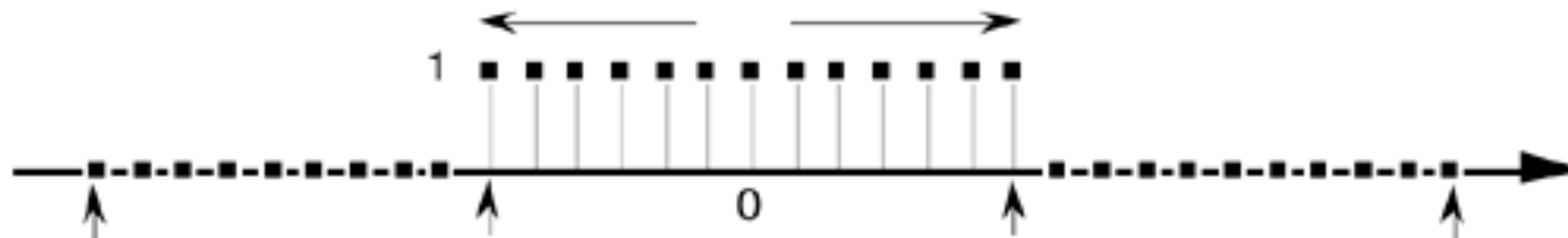
Frequency-Domain Response



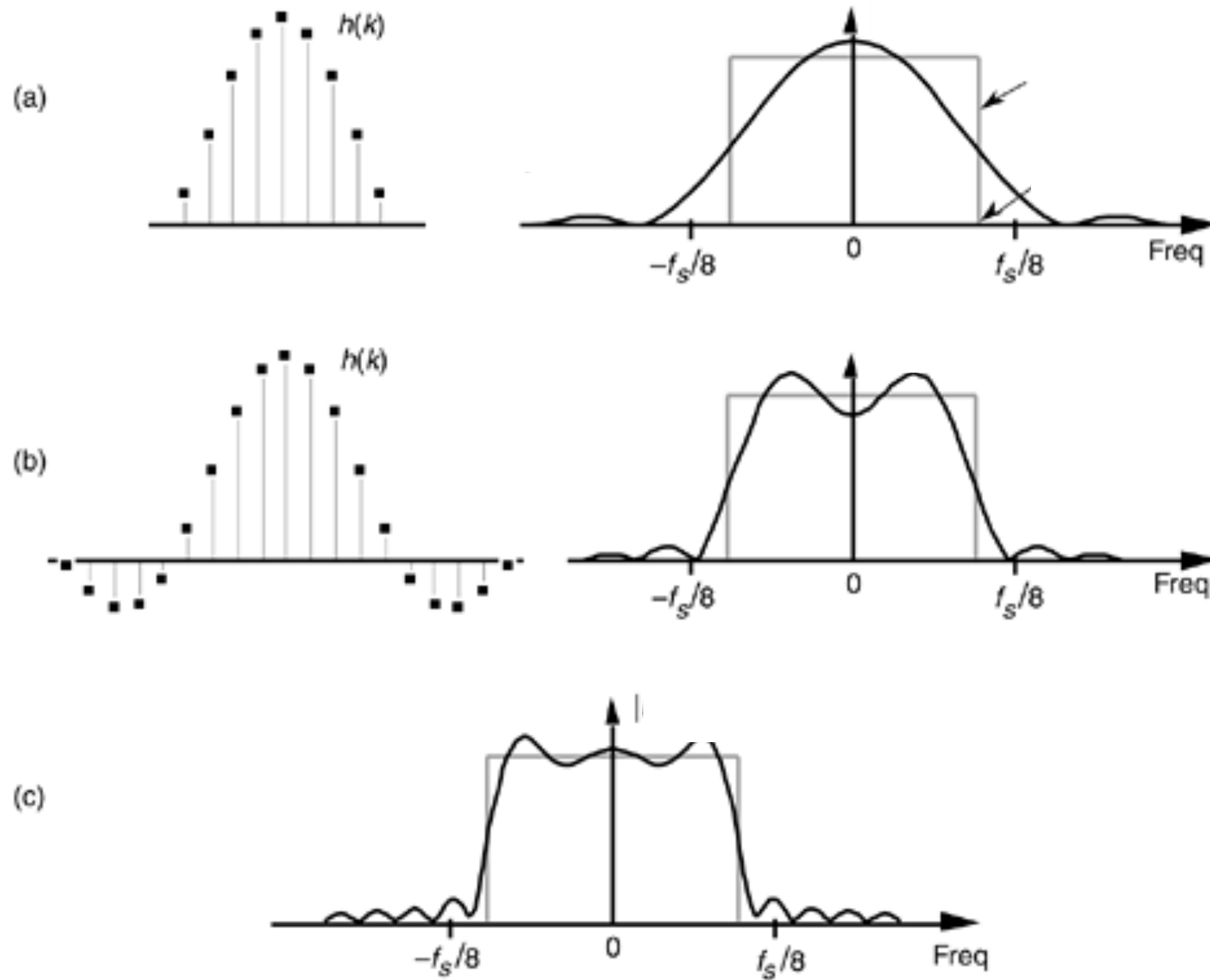
More FIR Filters



Ideal Low-Pass FIR Filter

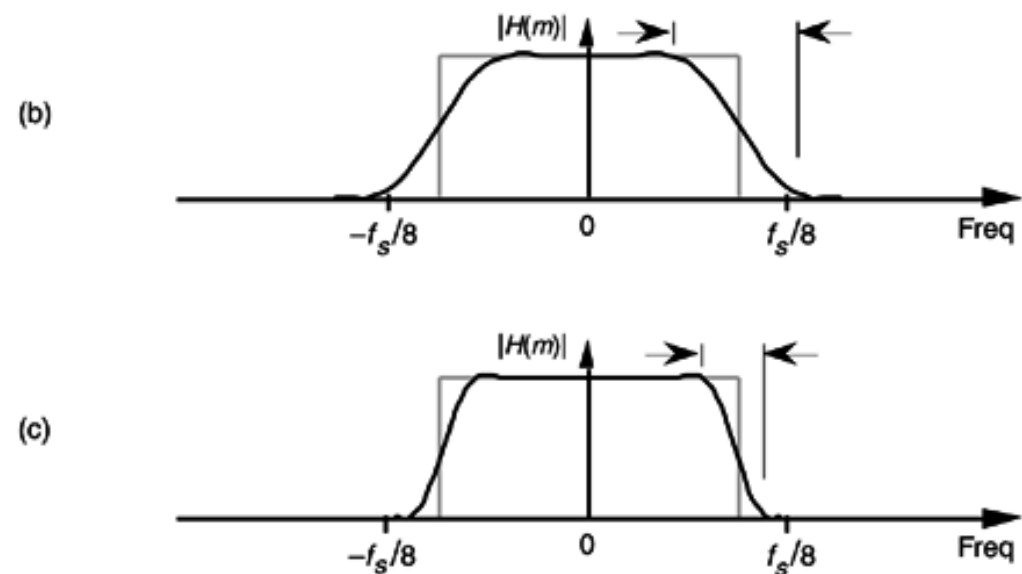
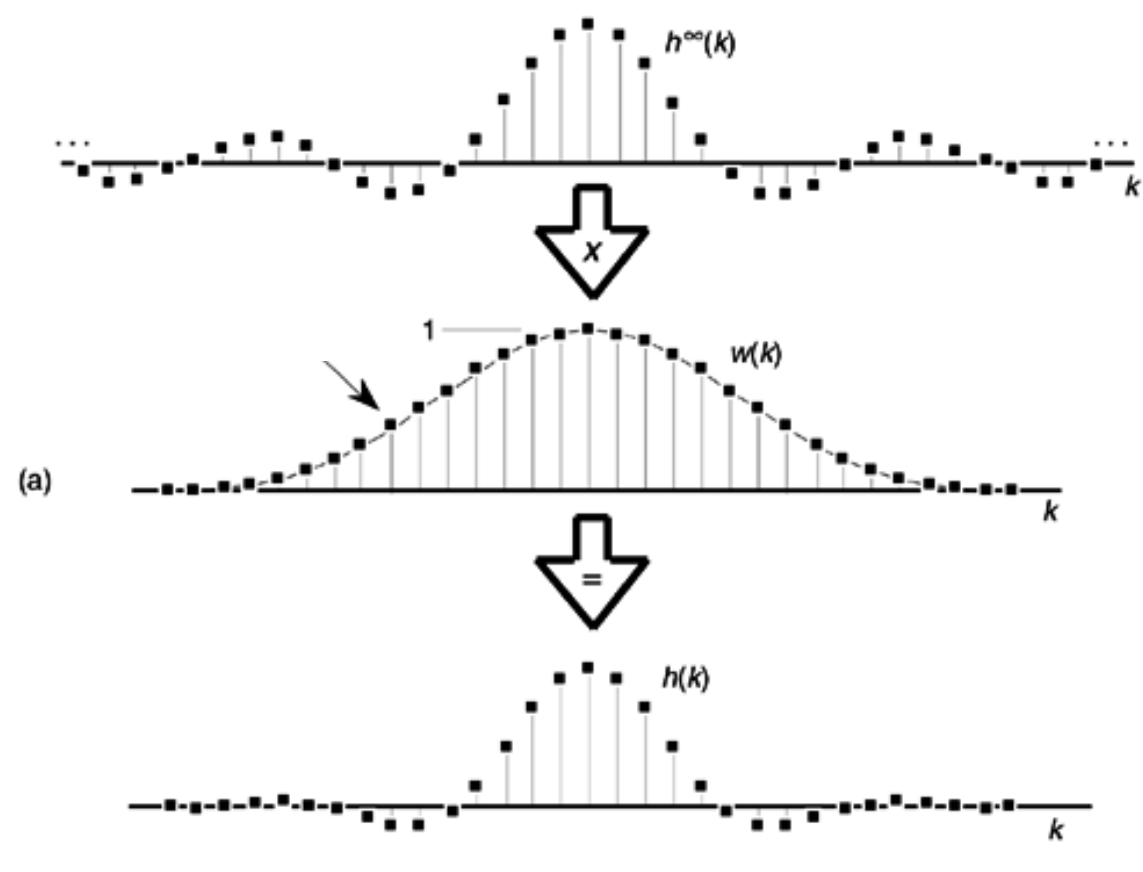


Convolved Low-Pass Filter

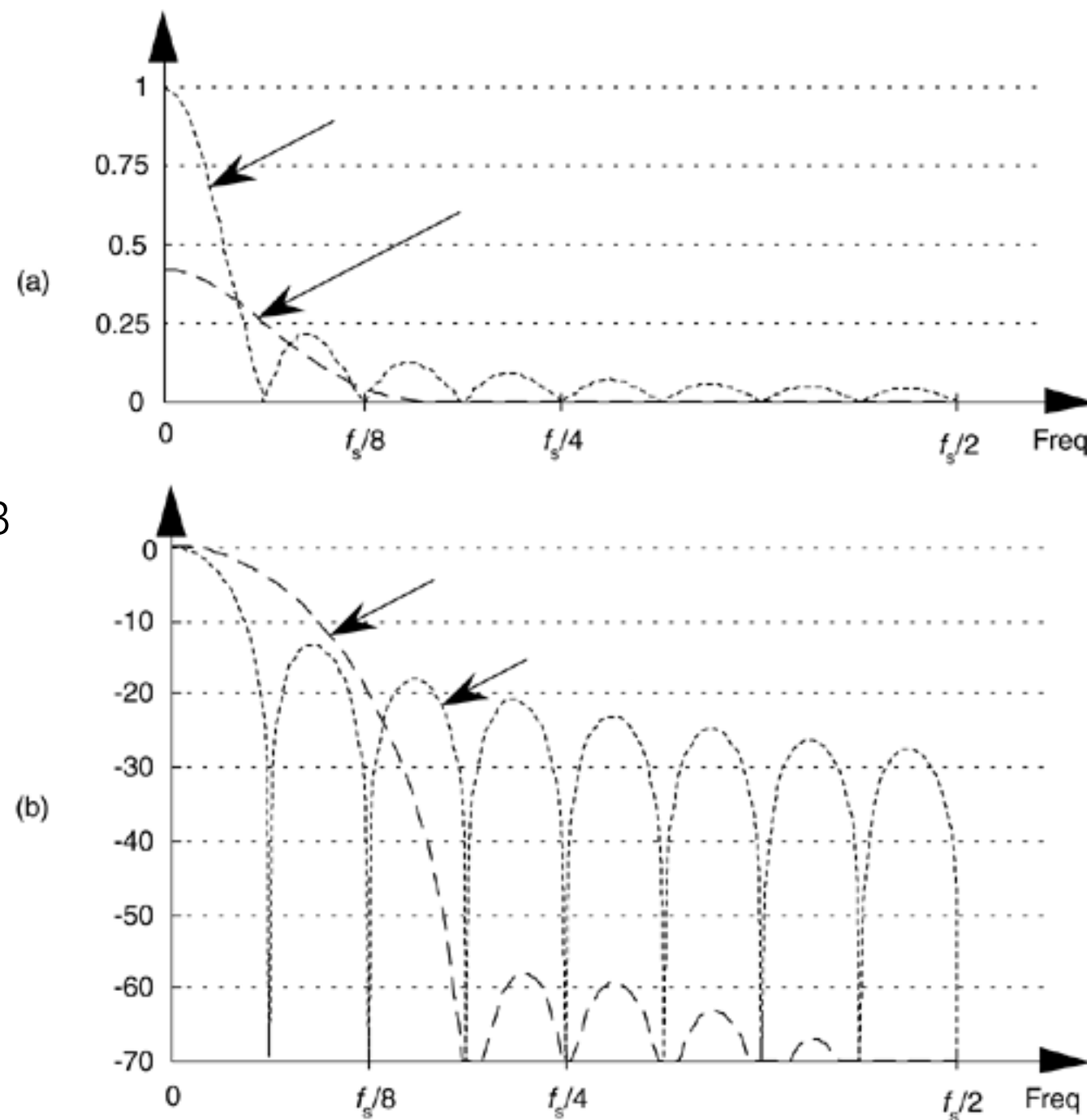


Example: Blackman Window

$$w(k) =$$



dB



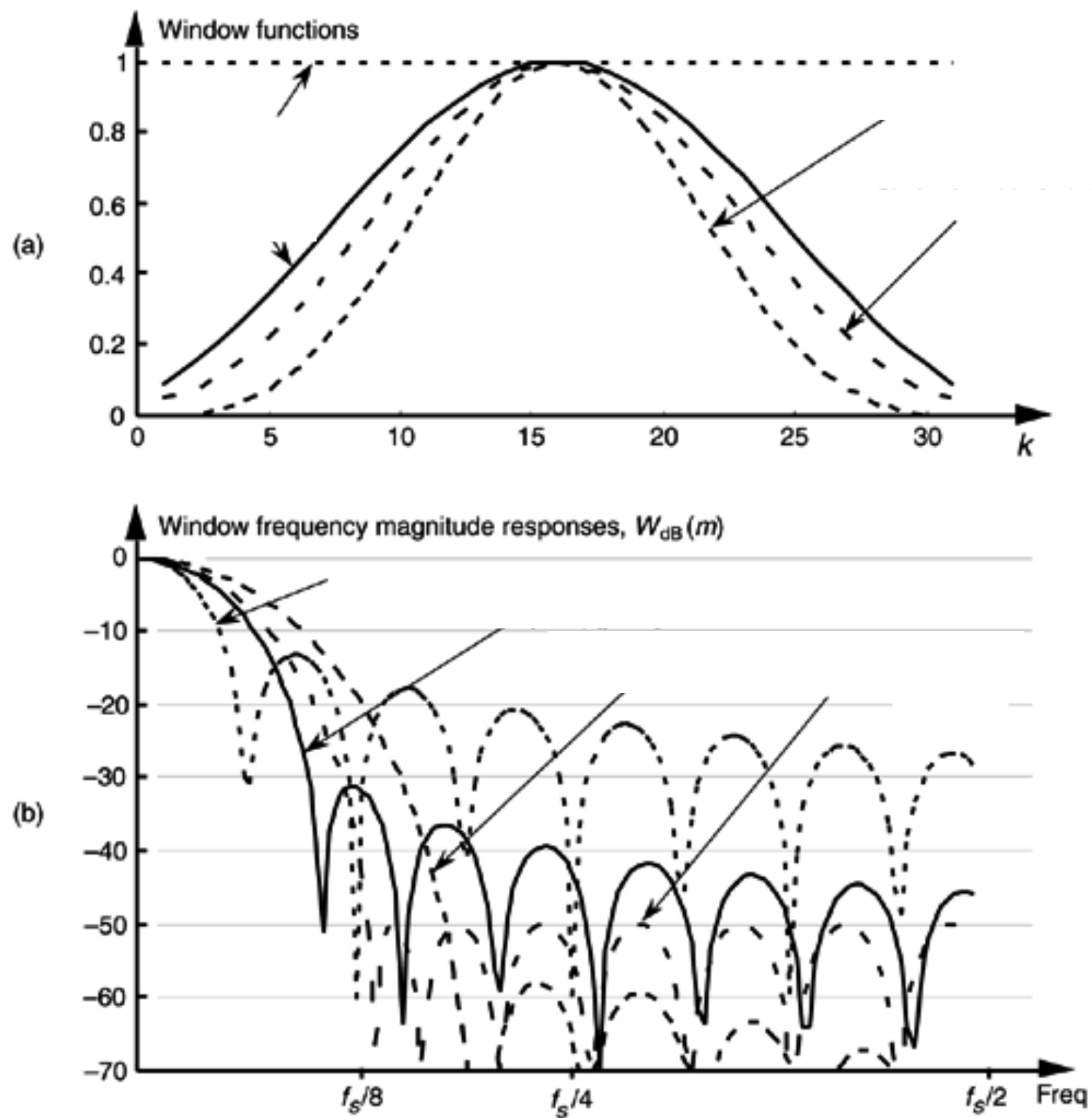
More (Tunable) Filters

- $w(k) =$

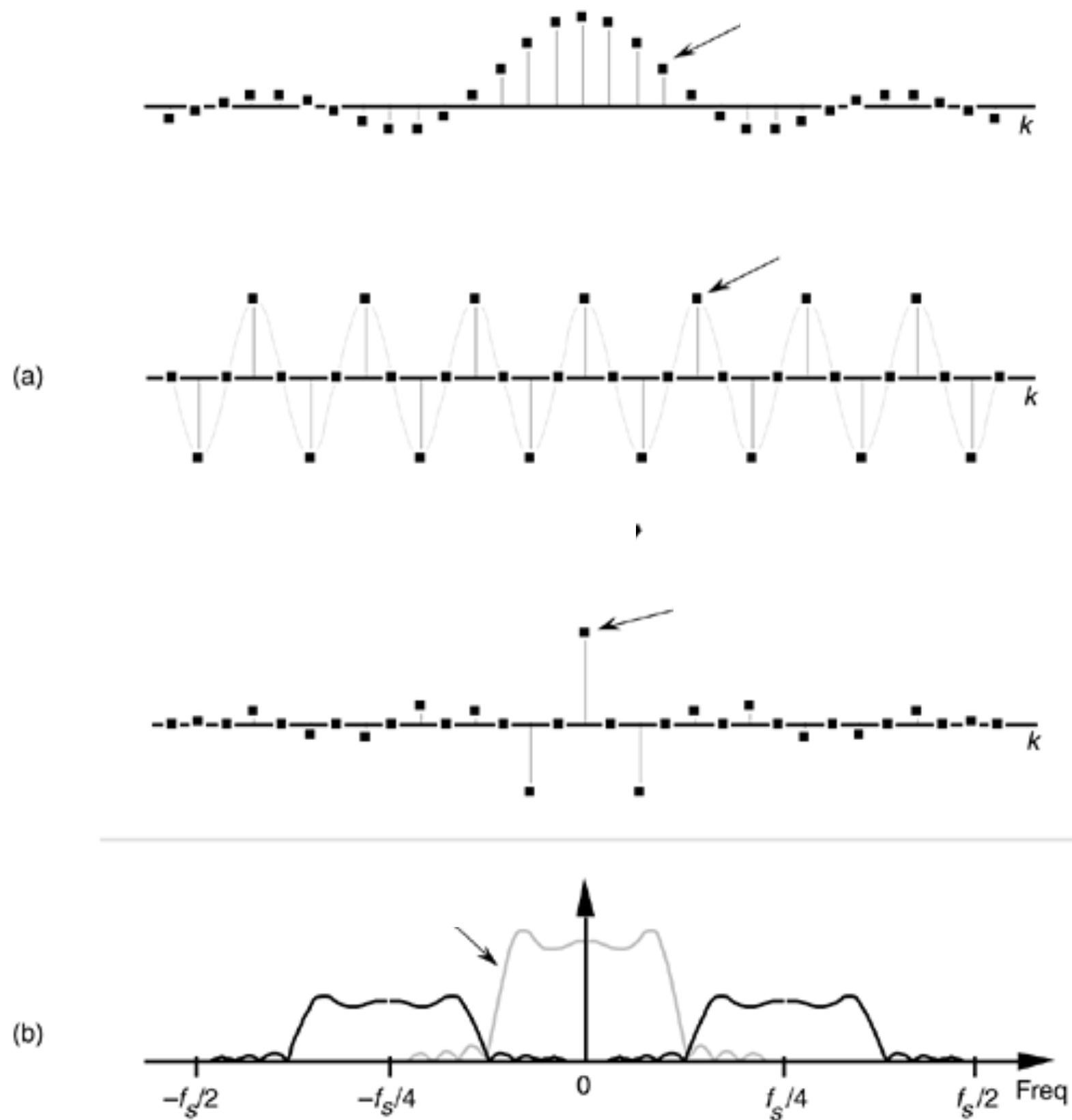
- where $\alpha =$

- $\omega(k) =$

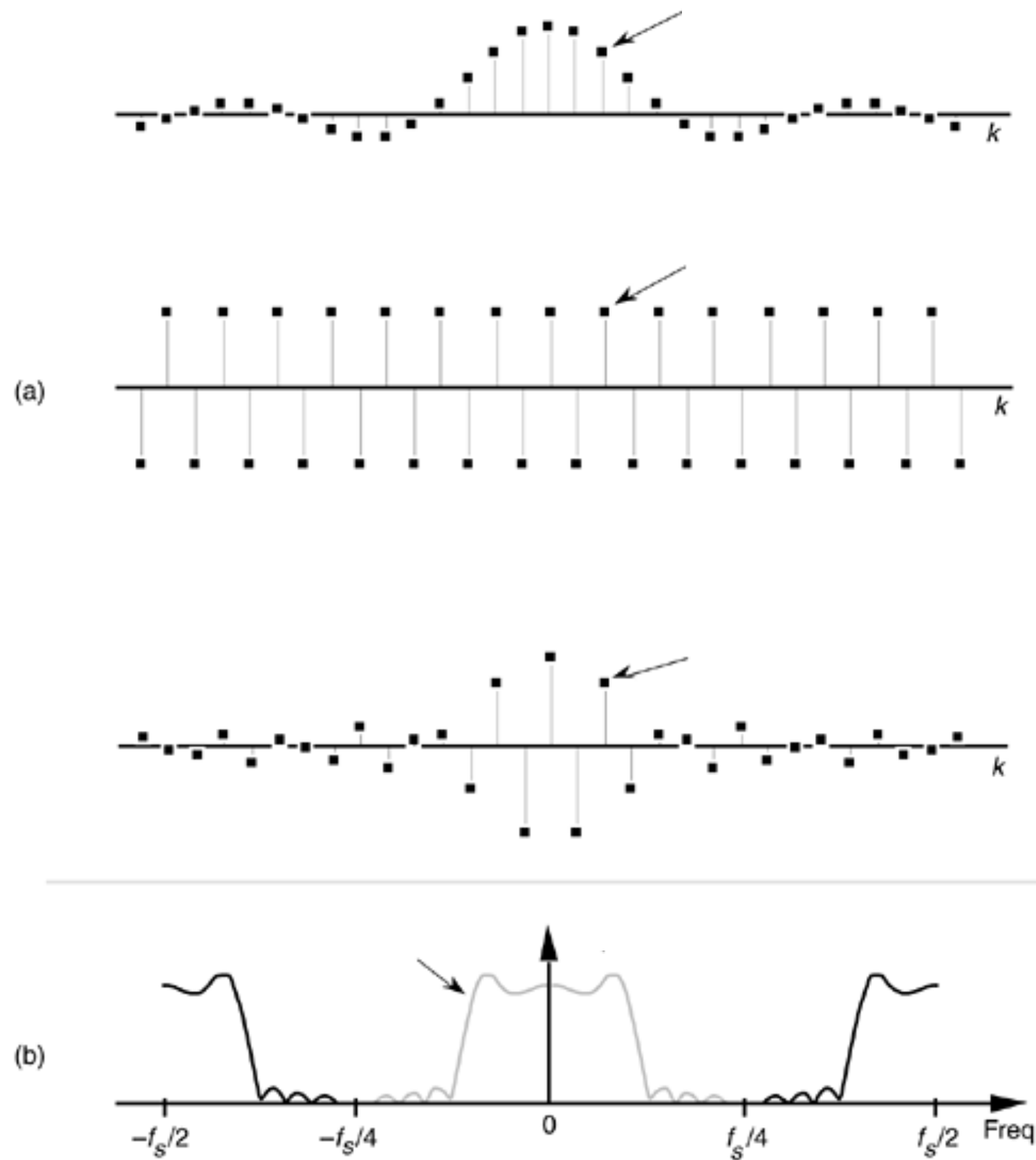
More Tunable Filters



Bandpass Filter

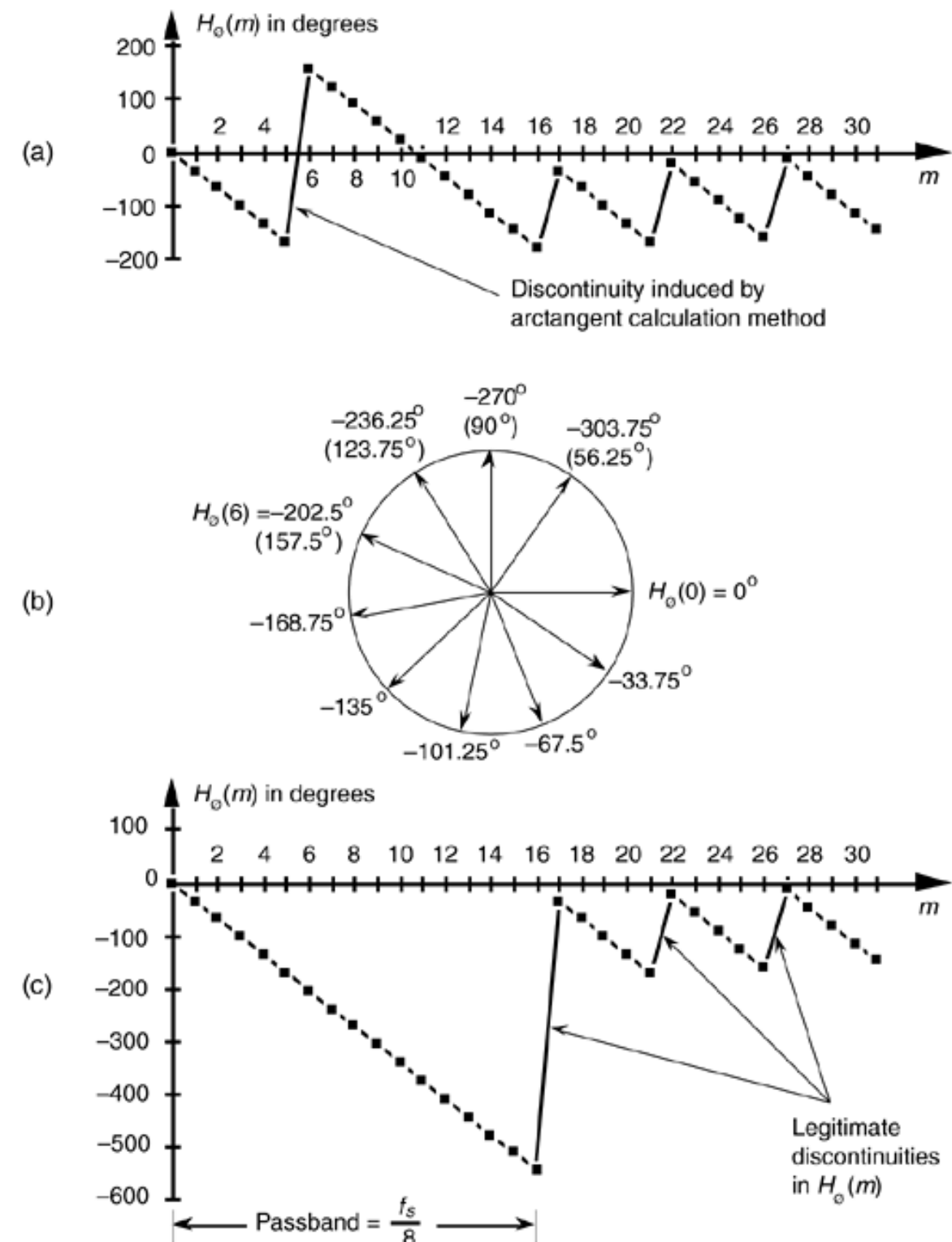
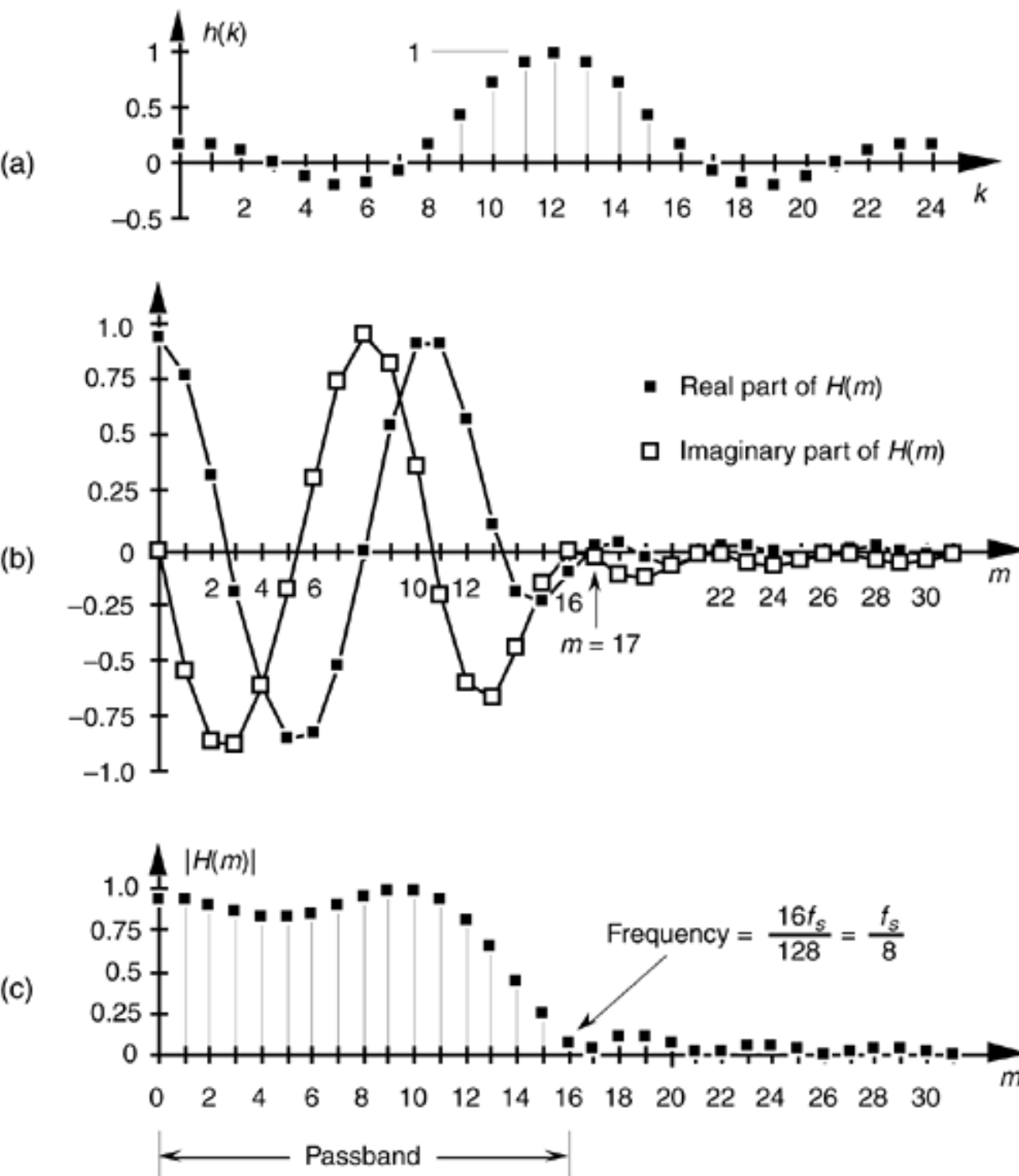


Highpass Filter

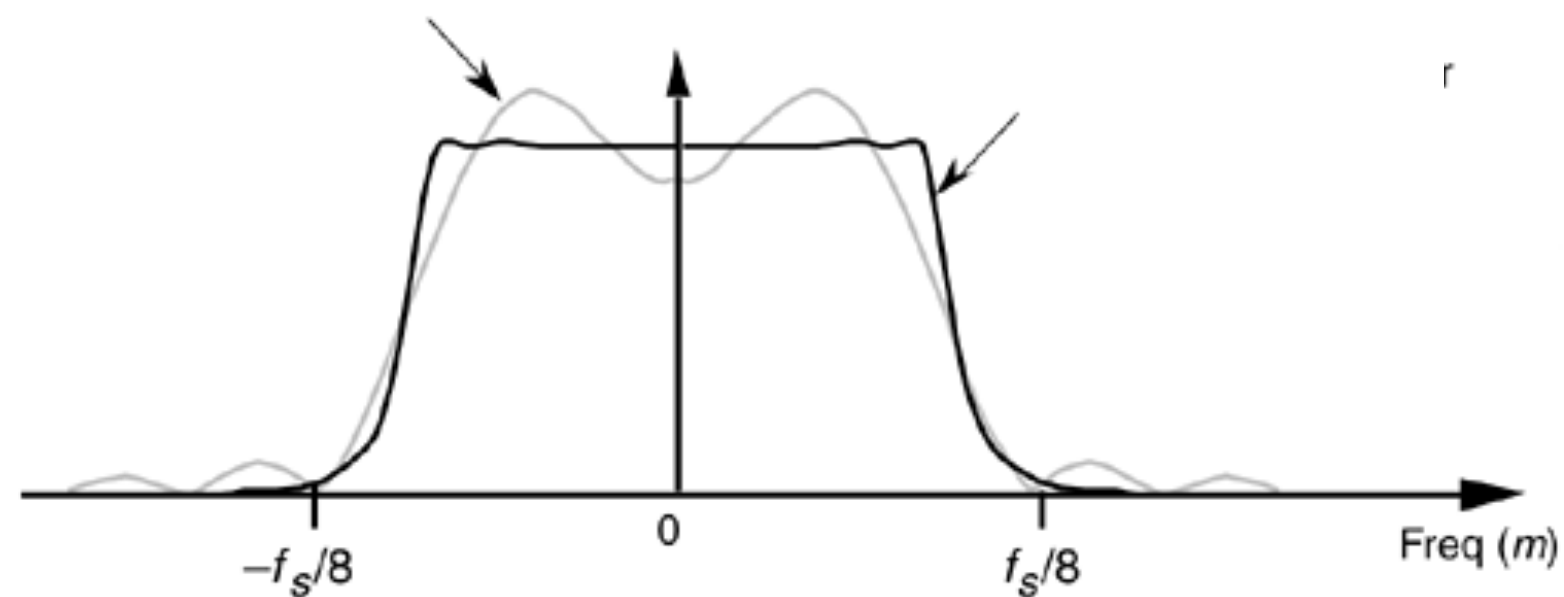
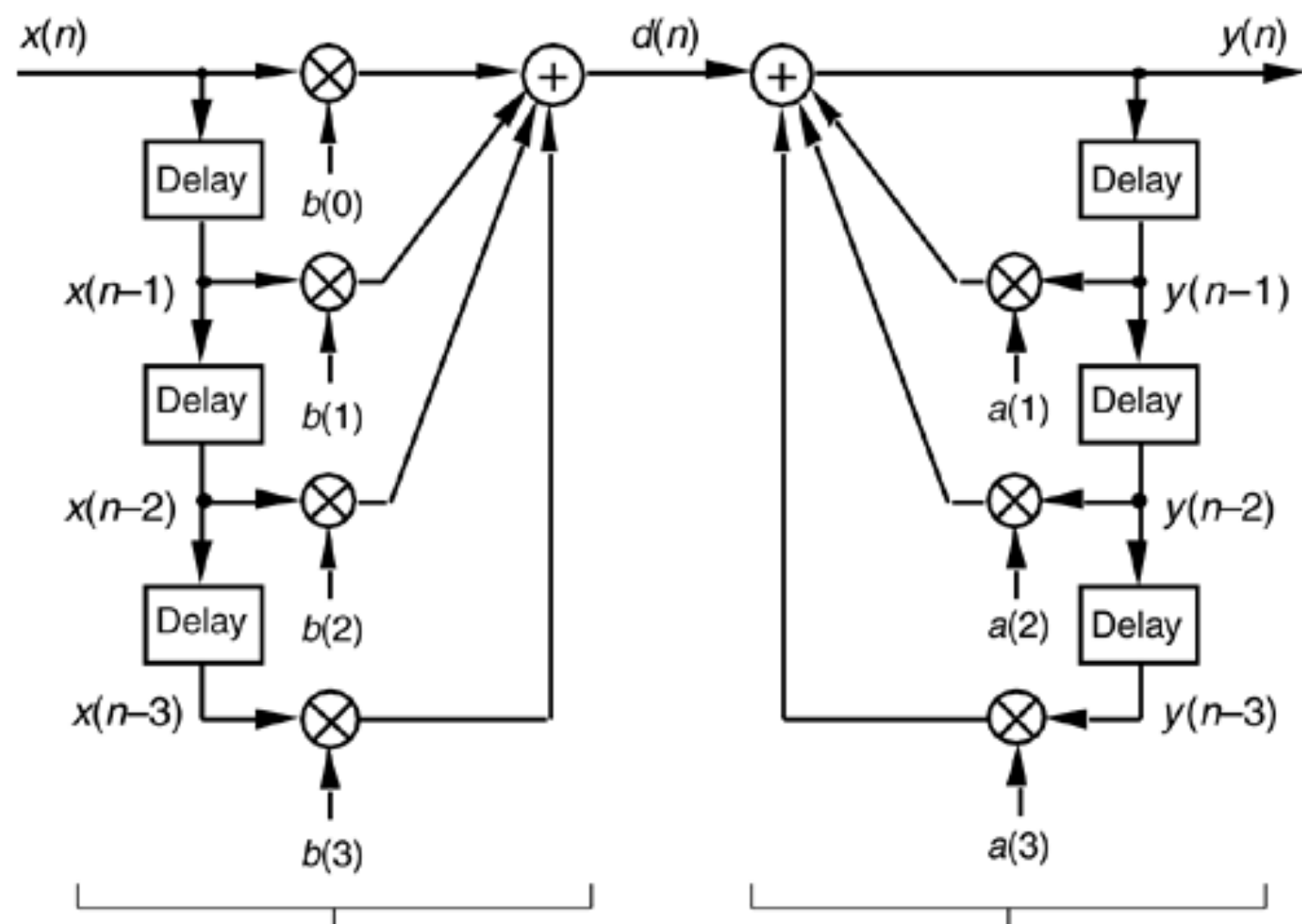


Phase Response in FIR Filters

phase shift in :
constant group delay (no):

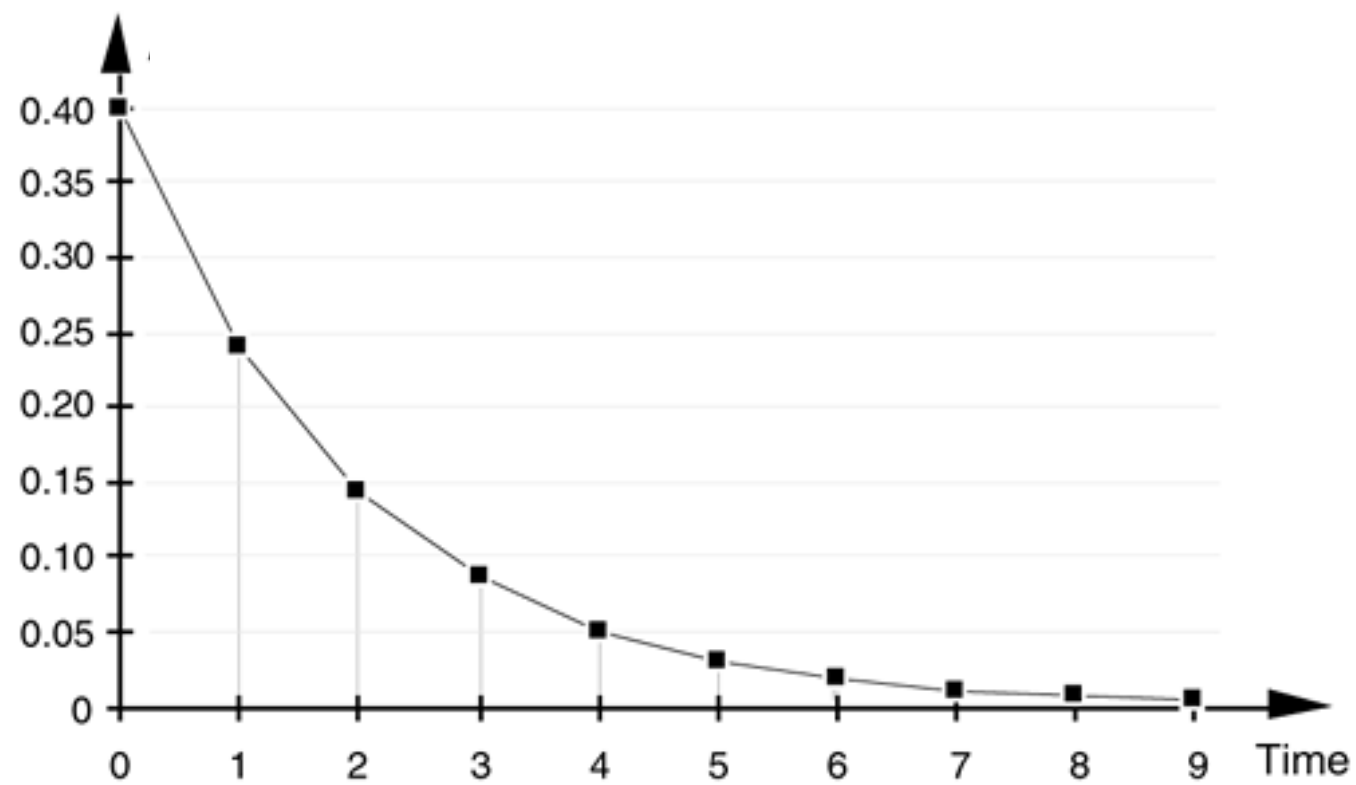
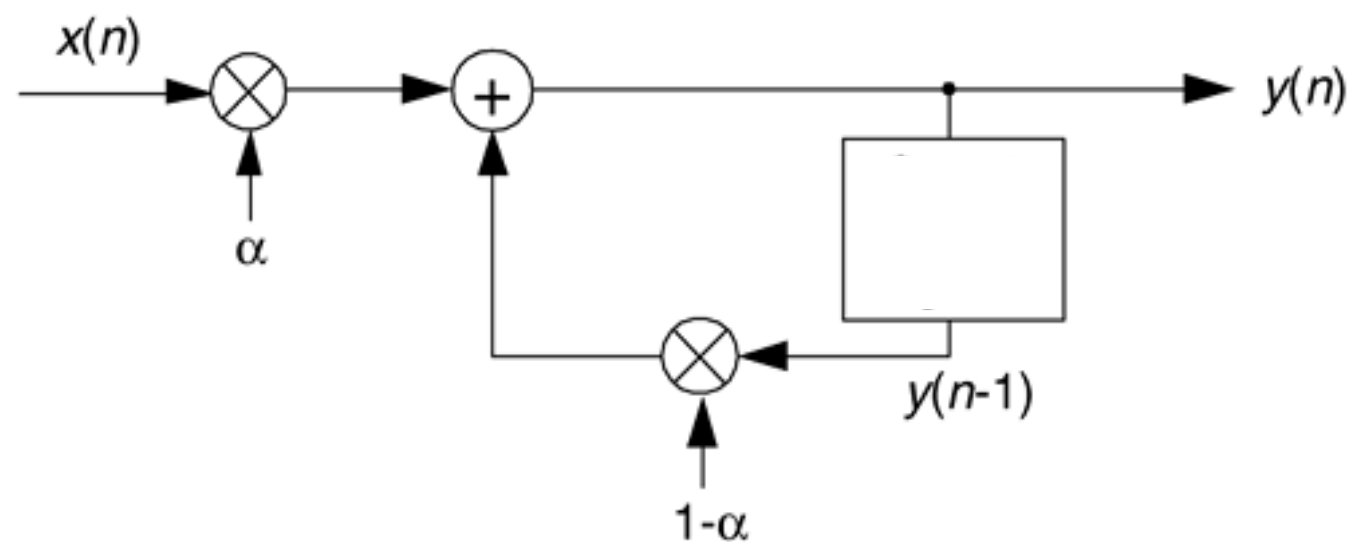


IIR Filters

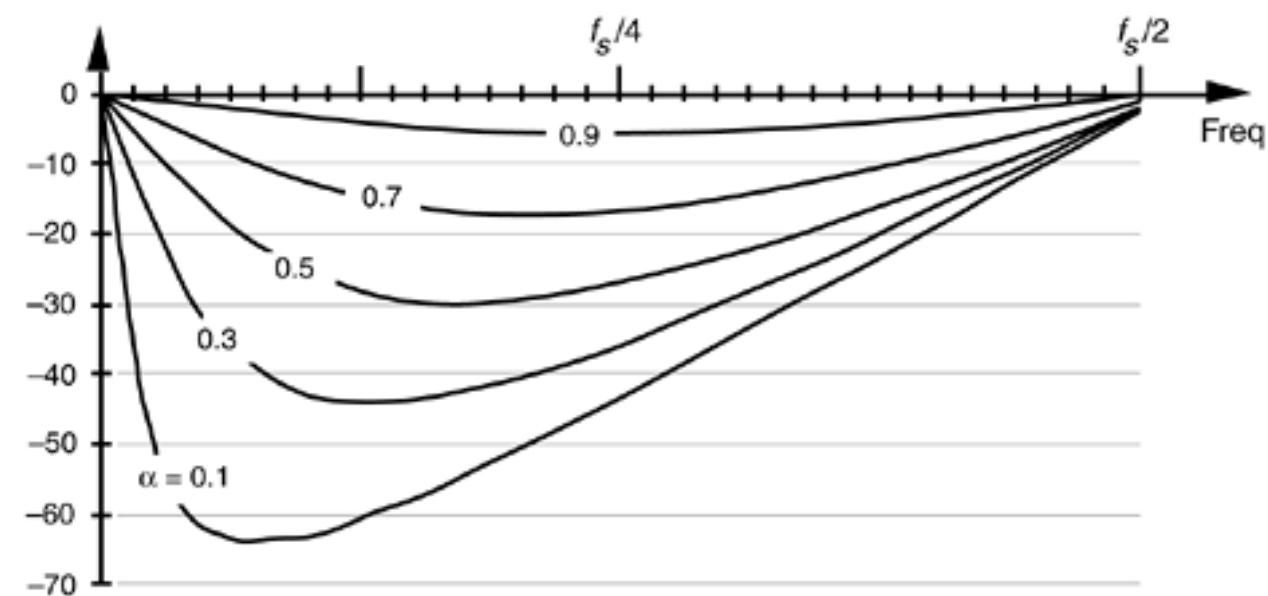
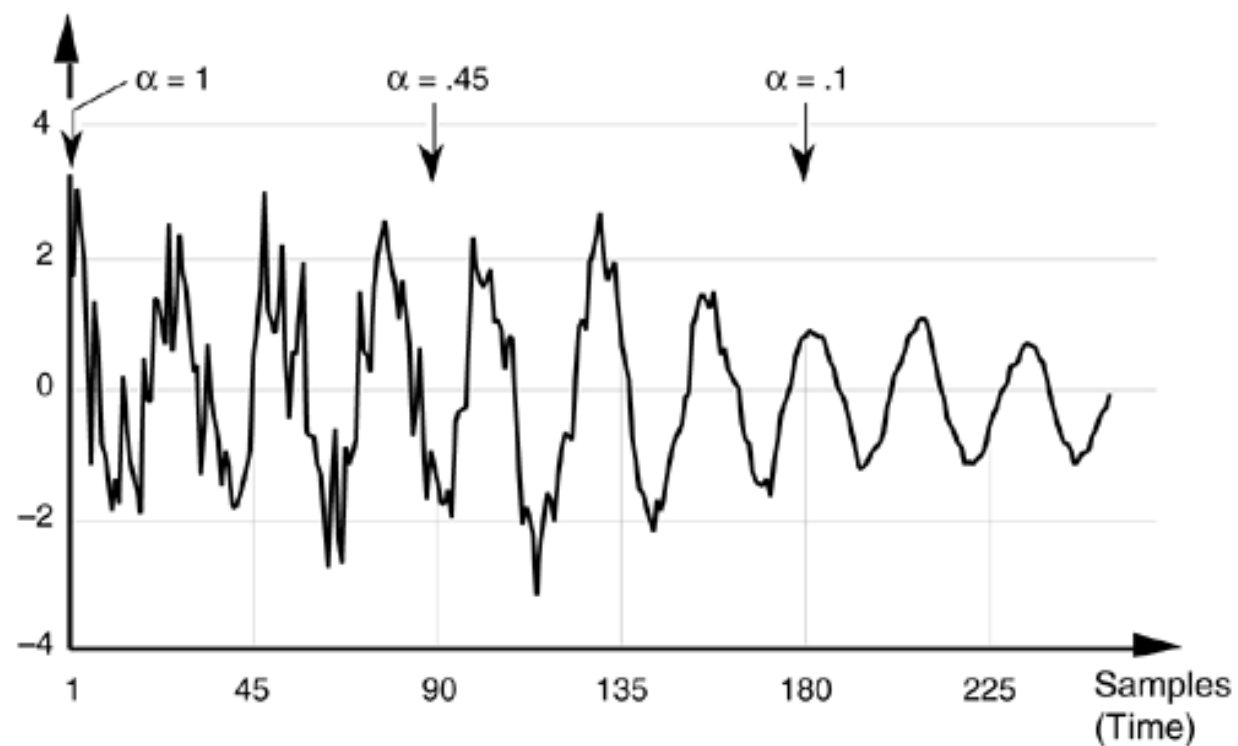
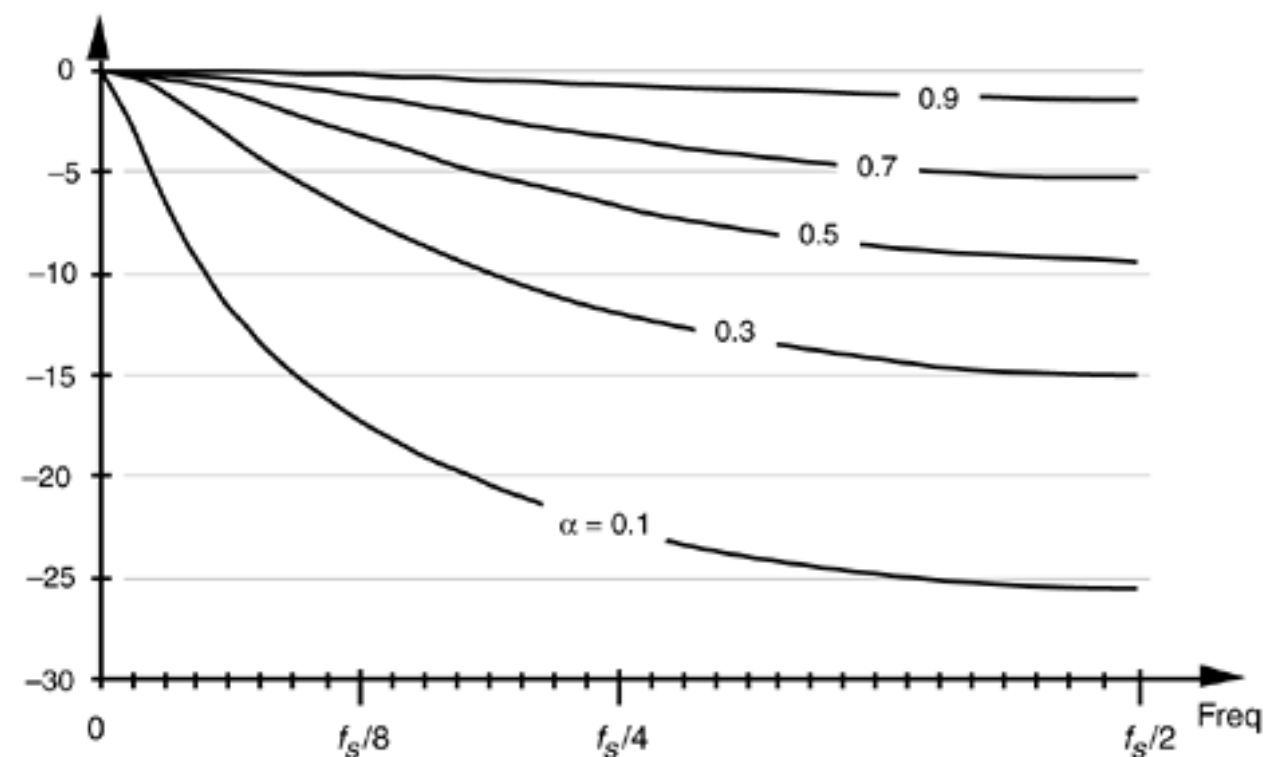
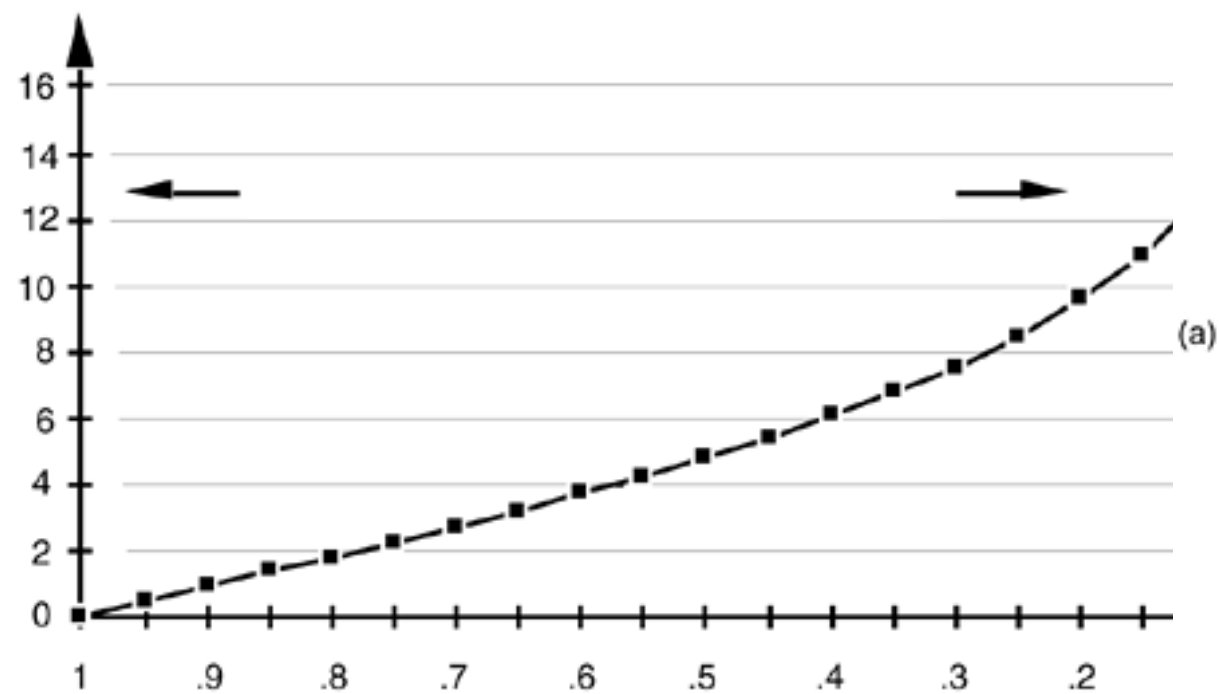


Example: Exponential Averaging Filter

$y(n) =$



Exponential IIR Filter



What We Have Not Covered

- Many topics to cover, so I so far focused on most immediately
 - There are topics dedicated to
- Other possible topics of interest (at your leisure):
 - Digital Signal Processing
 - Digital mixing
 - Modulation/demodulation
 - Smoothing, windowing
 - Often useful for image processing
 - Down-sampling (decimating), re-sampling