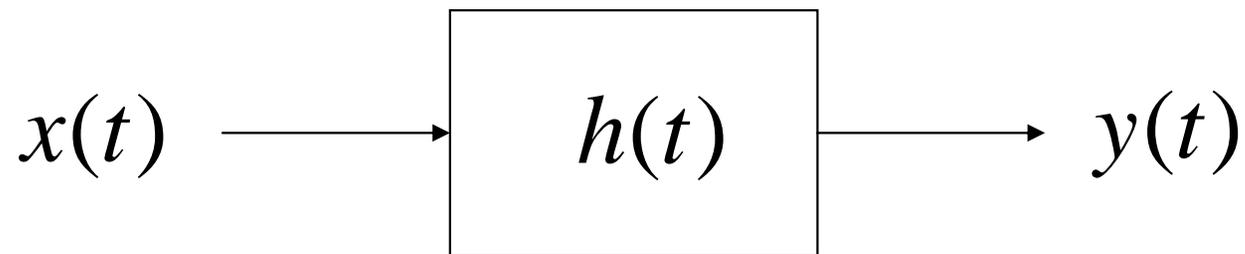


# Chapter 5

## Frequency Domain Analysis of Systems

# CT, LTI Systems

- Consider the following CT LTI system:



- Assumption: the impulse response  $h(t)$  is *absolutely integrable*, i.e.,

$$\int_{\mathbb{R}} |h(t)| dt < \infty$$

(this has to do with *system stability*)

# Response of a CT, LTI System to a Sinusoidal Input

- What's the response  $y(t)$  of this system to the input signal

$$x(t) = \cos(\omega_0 t + \theta), \quad t \in \mathbb{R} \quad ?$$

- We start by looking for the response  $y_c(t)$  of the same system to

$$x_c(t) = e^{j\omega_0 t} \quad t \in \mathbb{R}$$

# Response of a CT, LTI System to a Complex Exponential Input

- The output is obtained through convolution as

$$\begin{aligned}y_c(t) &= h(t) * x_c(t) = \int_{\mathbb{R}} h(\tau) x_c(t - \tau) d\tau = \\&= \int_{\mathbb{R}} h(\tau) e^{j\omega_0(t-\tau)} d\tau = \\&= \underbrace{e^{j\omega_0 t}}_{x_c(t)} \int_{\mathbb{R}} h(\tau) e^{-j\omega_0 \tau} d\tau = \\&= x_c(t) \int_{\mathbb{R}} h(\tau) e^{-j\omega_0 \tau} d\tau\end{aligned}$$

# The Frequency Response of a CT, LTI System

- By defining

$$H(\omega) = \int_{\mathbb{R}} h(\tau) e^{-j\omega\tau} d\tau$$

$H(\omega)$  is the frequency response of the CT, LTI system = Fourier transform of  $h(t)$

it is

$$\begin{aligned} y_c(t) &= H(\omega_0) x_c(t) = \\ &= H(\omega_0) e^{j\omega_0 t}, \quad t \in \mathbb{R} \end{aligned}$$

- Therefore, the response of the LTI system to a complex exponential is another complex exponential with the same frequency  $\omega_0$

# Analyzing the Output Signal $y_c(t)$

- Since  $H(\omega_0)$  is in general a complex quantity, we can write

$$\begin{aligned} y_c(t) &= H(\omega_0)e^{j\omega_0 t} = \\ &= |H(\omega_0)| e^{j \arg H(\omega_0)} e^{j\omega_0 t} = \\ &= \underbrace{|H(\omega_0)|}_{\text{output signal's magnitude}} e^{j(\omega_0 t + \underbrace{\arg H(\omega_0)}_{\text{output signal's phase}})} \end{aligned}$$

# Response of a CT, LTI System to a Sinusoidal Input

- With Euler's formulas we can express  $x(t)$  as

$$\begin{aligned}x(t) &= \cos(\omega_0 t + \theta) \\ &= \frac{1}{2} (e^{j(\omega_0 t + \theta)} + e^{-j(\omega_0 t + \theta)}) \\ &= \frac{1}{2} e^{j\theta} e^{j\omega_0 t} + \frac{1}{2} e^{-j\theta} e^{-j\omega_0 t}\end{aligned}$$

Using the previous result, the response is

$$y(t) = \frac{1}{2} e^{j\theta} H(\omega_0) e^{j\omega_0 t} + \frac{1}{2} e^{-j\theta} H(-\omega_0) e^{-j\omega_0 t}$$

# Response of a CT, LTI System to a Sinusoidal Input – Cont'd

- If  $h(t)$  is real, then  $H(-\omega) = H^*(\omega)$  and

$$H(\omega_0) = |H(\omega_0)| e^{j \arg H(\omega_0)}$$

$$H(-\omega_0) = |H(\omega_0)| e^{-j \arg H(\omega_0)}$$

- Thus we can write  $y(t)$  as

$$\begin{aligned} y(t) &= \frac{1}{2} |H(\omega_0)| e^{j(\omega_0 t + \theta + \arg H(\omega_0))} + \frac{1}{2} |H(\omega_0)| e^{-j(\omega_0 t + \theta + \arg H(\omega_0))} \\ &= |H(\omega_0)| \cos(\omega_0 t + \theta + \arg H(\omega_0)) \end{aligned}$$

# Response of a CT, LTI System to a Sinusoidal Input – Cont'd

- Thus, the response to

$$x(t) = A \cos(\omega_0 t + \theta)$$

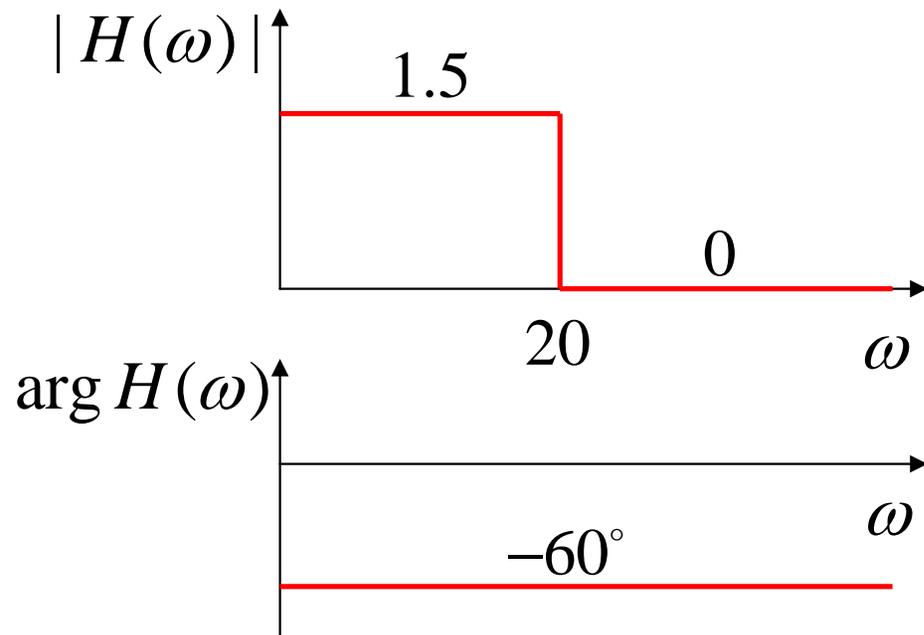
is

$$y(t) = A |H(\omega_0)| \cos(\omega_0 t + \theta + \arg H(\omega_0))$$

which is also a sinusoid with the same frequency  $\omega_0$  but with the **amplitude scaled by the factor  $|H(\omega_0)|$**  and with the **phase shifted by amount  $\arg H(\omega_0)$**

# Example: Response of a CT, LTI System to Sinusoidal Inputs

- Suppose that the frequency response of a CT, LTI system is defined by the following specs:



$$|H(\omega)| = \begin{cases} 1.5, & 0 \leq \omega \leq 20, \\ 0, & \omega > 20, \end{cases}$$

$$\arg H(\omega) = -60^\circ, \forall \omega$$

## Example: Response of a CT, LTI System to Sinusoidal Inputs – Cont'd

- If the input to the system is

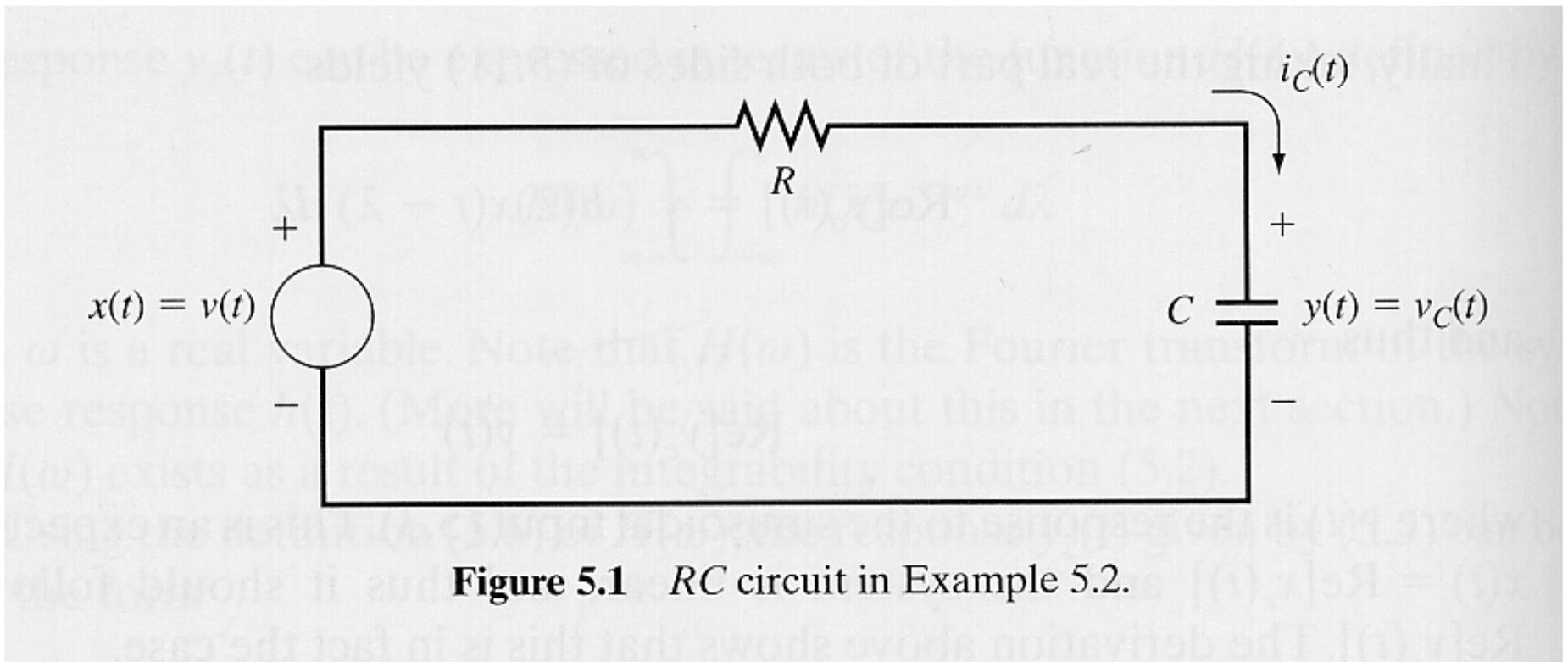
$$x(t) = 2 \cos(10t + 90^\circ) + 5 \cos(25t + 120^\circ)$$

- Then the output is

$$\begin{aligned} y(t) &= 2 |H(10)| \cos(10t + 90^\circ + \arg H(10)) + \\ &\quad + 5 |H(25)| \cos(25t + 120^\circ + \arg H(25)) = \\ &= 3 \cos(10t + 30^\circ) \end{aligned}$$

# Example: Frequency Analysis of an RC Circuit

- Consider the RC circuit shown in figure



**Figure 5.1** RC circuit in Example 5.2.

## Example: Frequency Analysis of an RC Circuit – Cont'd

- From EEE2032F, we know that:
  1. The **complex impedance** of the capacitor is equal to  $1/j\omega C$
  2. If the input voltage is  $x_c(t) = e^{j\omega t}$ , then the output signal is given by

$$y_c(t) = \frac{1/j\omega C}{R + 1/j\omega C} e^{j\omega t} = \frac{1/RC}{j\omega + 1/RC} e^{j\omega t}$$

## Example: Frequency Analysis of an RC Circuit – Cont'd

- Setting  $\omega = \omega_0$ , it is

$$x_c(t) = e^{j\omega_0 t} \quad \text{and} \quad y_c(t) = \frac{1/RC}{j\omega_0 + 1/RC} e^{j\omega_0 t}$$

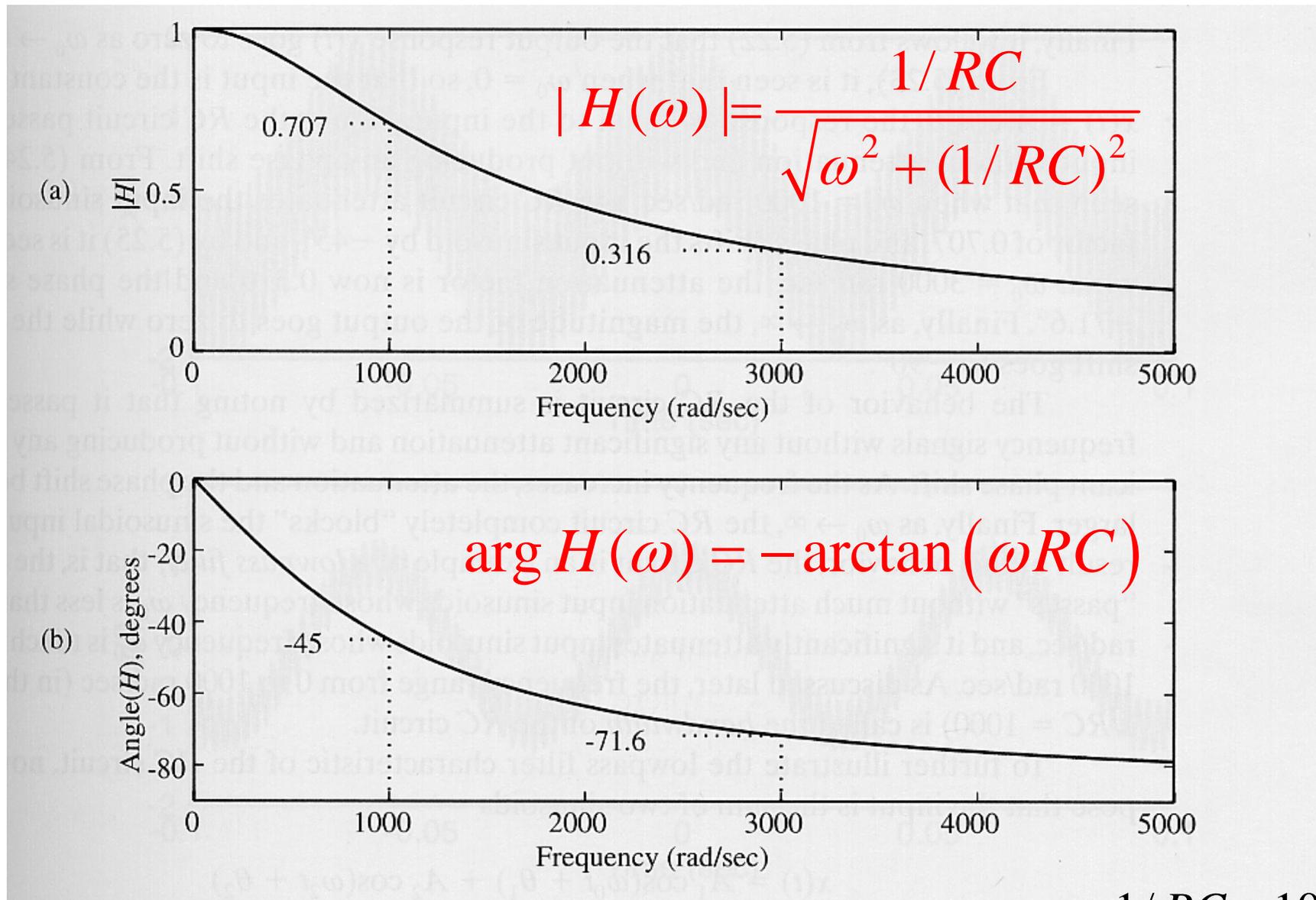
whence we can write

$$y_c(t) = H(\omega_0)x_c(t)$$

where

$$H(\omega) = \frac{1/RC}{j\omega + 1/RC}$$

# Example: Frequency Analysis of an RC Circuit – Cont'd



$1/RC = 1000$

## Example: Frequency Analysis of an RC Circuit – Cont'd

- The knowledge of the frequency response  $H(\omega)$  allows us to compute the response  $y(t)$  of the system to any sinusoidal input signal

$$x(t) = A \cos(\omega_0 t + \theta)$$

since

$$y(t) = A |H(\omega_0)| \cos(\omega_0 t + \theta + \arg H(\omega_0))$$

## Example: Frequency Analysis of an RC Circuit – Cont'd

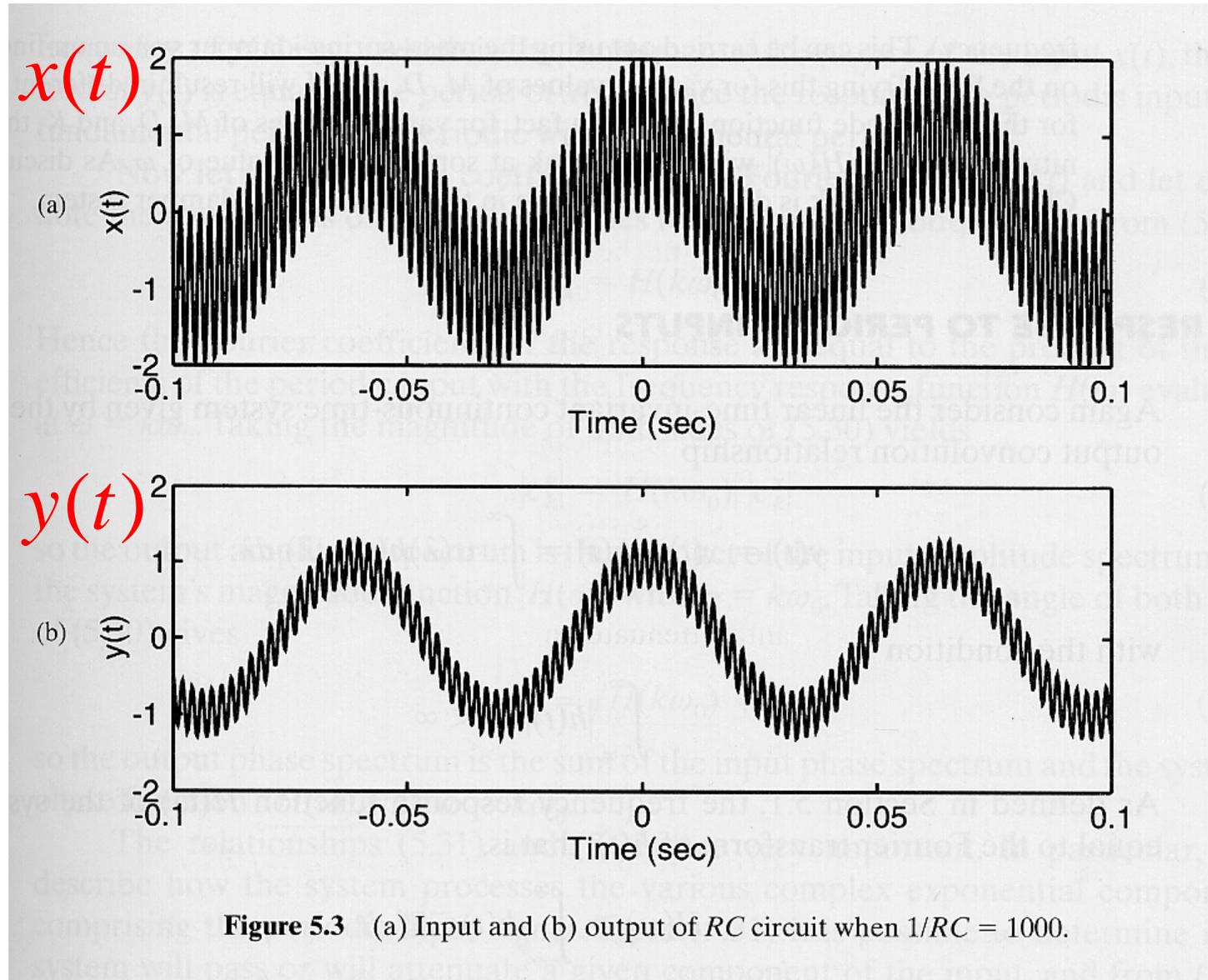
- Suppose that  $1/RC = 1000$  and that

$$x(t) = \cos(100t) + \cos(3000t)$$

- Then, the output signal is

$$\begin{aligned} y(t) &= |H(100)| \cos(100t + \arg H(100)) + \\ &+ |H(3000)| \cos(3000t + \arg H(3000)) = \\ &= 0.9950 \cos(100t - 5.71^\circ) + 0.3162 \cos(3000t - 71.56^\circ) \end{aligned}$$

# Example: Frequency Analysis of an RC Circuit – Cont'd



## Example: Frequency Analysis of an RC Circuit – Cont'd

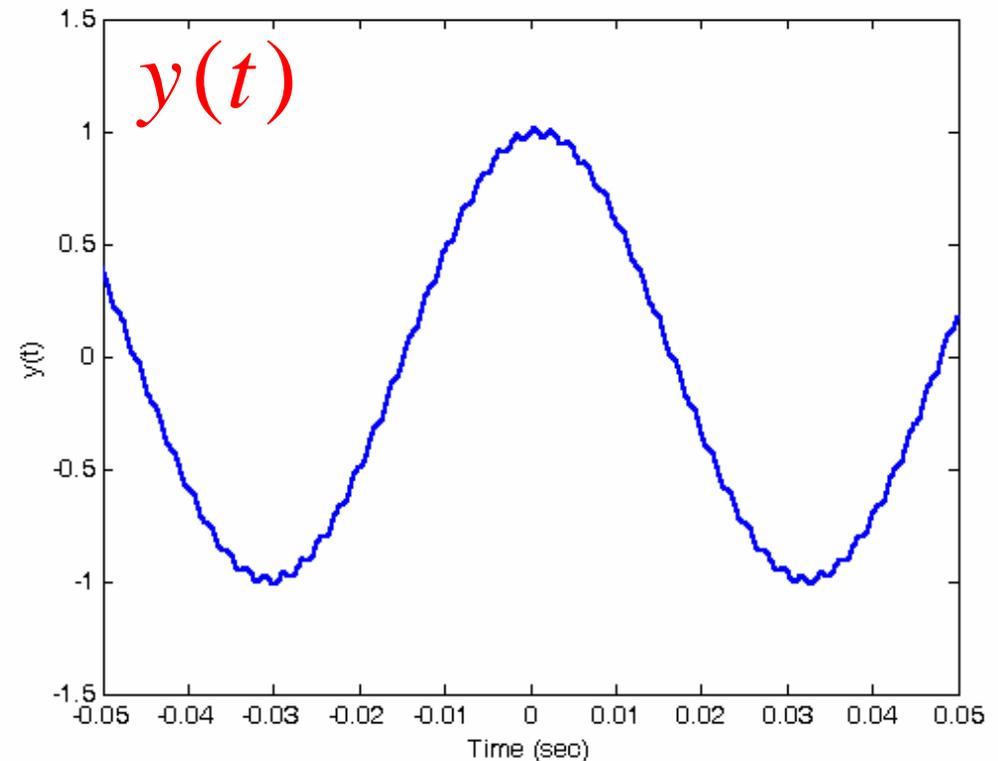
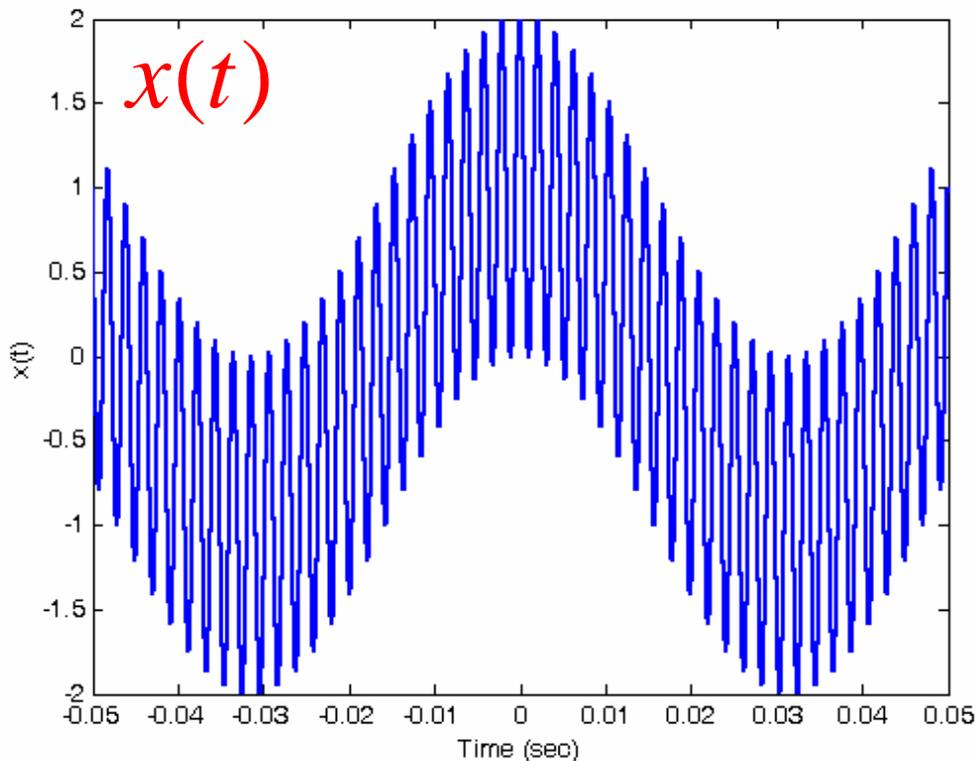
- Suppose now that

$$x(t) = \cos(100t) + \cos(50,000t)$$

- Then, the output signal is

$$\begin{aligned} y(t) &= |H(100)| \cos(100t + \arg H(100)) + \\ &+ |H(50,000)| \cos(50,000t + \arg H(50,000)) = \\ &= 0.9950 \cos(100t - 5.71^\circ) + 0.0200 \cos(50,000t - 88.85^\circ) \end{aligned}$$

# Example: Frequency Analysis of an RC Circuit – Cont'd



The RC circuit behaves as a **lowpass filter**, by letting low-frequency sinusoidal signals pass with little attenuation and by significantly attenuating high-frequency sinusoidal signals

# Response of a CT, LTI System to Periodic Inputs

- Suppose that the input to the CT, LTI system is a **periodic signal**  $x(t)$  having period  $T$
- This signal can be represented through its **Fourier series** as

$$x(t) = \sum_{k=-\infty}^{\infty} c_k^x e^{jk\omega_0 t}, \quad t \in \mathbb{R}$$

where

$$c_k^x = \frac{1}{T} \int_{t_0}^{t_0+T} x(t) e^{-jk\omega_0 t} dt, \quad k \in \mathbb{Z}$$

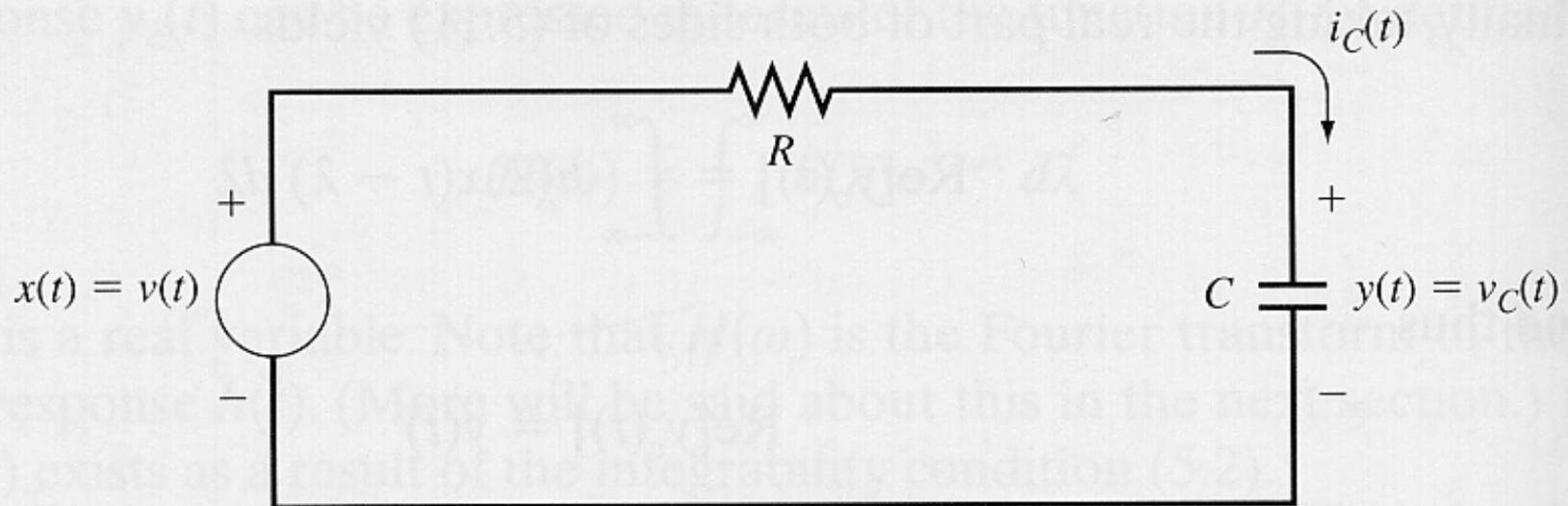
# Response of a CT, LTI System to Periodic Inputs – Cont'd

- By exploiting the previous results and the linearity of the system, the output of the system is

$$\begin{aligned} y(t) &= \sum_{k=-\infty}^{\infty} H(k\omega_0) c_k^x e^{jk\omega_0 t} \\ &= \sum_{k=-\infty}^{\infty} \underbrace{|H(k\omega_0)| |c_k^x|}_{|c_k^y|} e^{j(k\omega_0 t + \underbrace{\arg(c_k^x) + \arg H(k\omega_0)}_{\arg c_k^y})} = \\ &= \sum_{k=-\infty}^{\infty} |c_k^y| e^{j(k\omega_0 t + \arg(c_k^y))} = \sum_{k=-\infty}^{\infty} c_k^y e^{jk\omega_0 t}, \quad t \in \mathbb{R} \end{aligned}$$

# Example: Response of an RC Circuit to a Rectangular Pulse Train

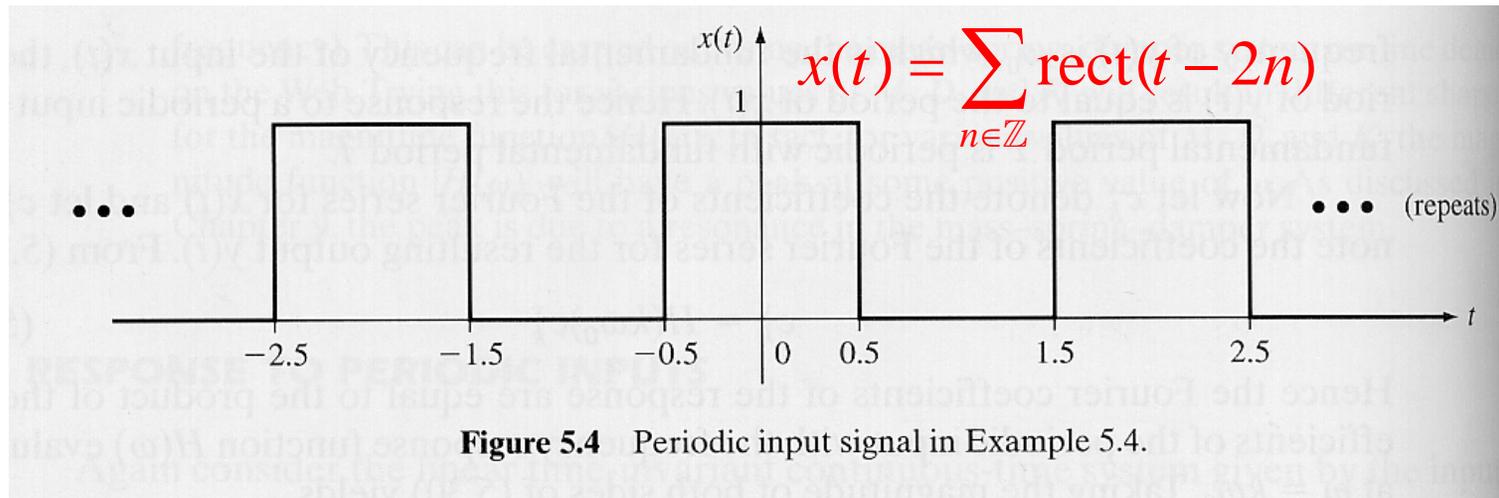
- Consider the RC circuit



**Figure 5.1** RC circuit in Example 5.2.

with input 
$$x(t) = \sum_{n \in \mathbb{Z}} \text{rect}(t - 2n)$$

# Example: Response of an RC Circuit to a Rectangular Pulse Train – Cont'd



- We have found its Fourier series to be

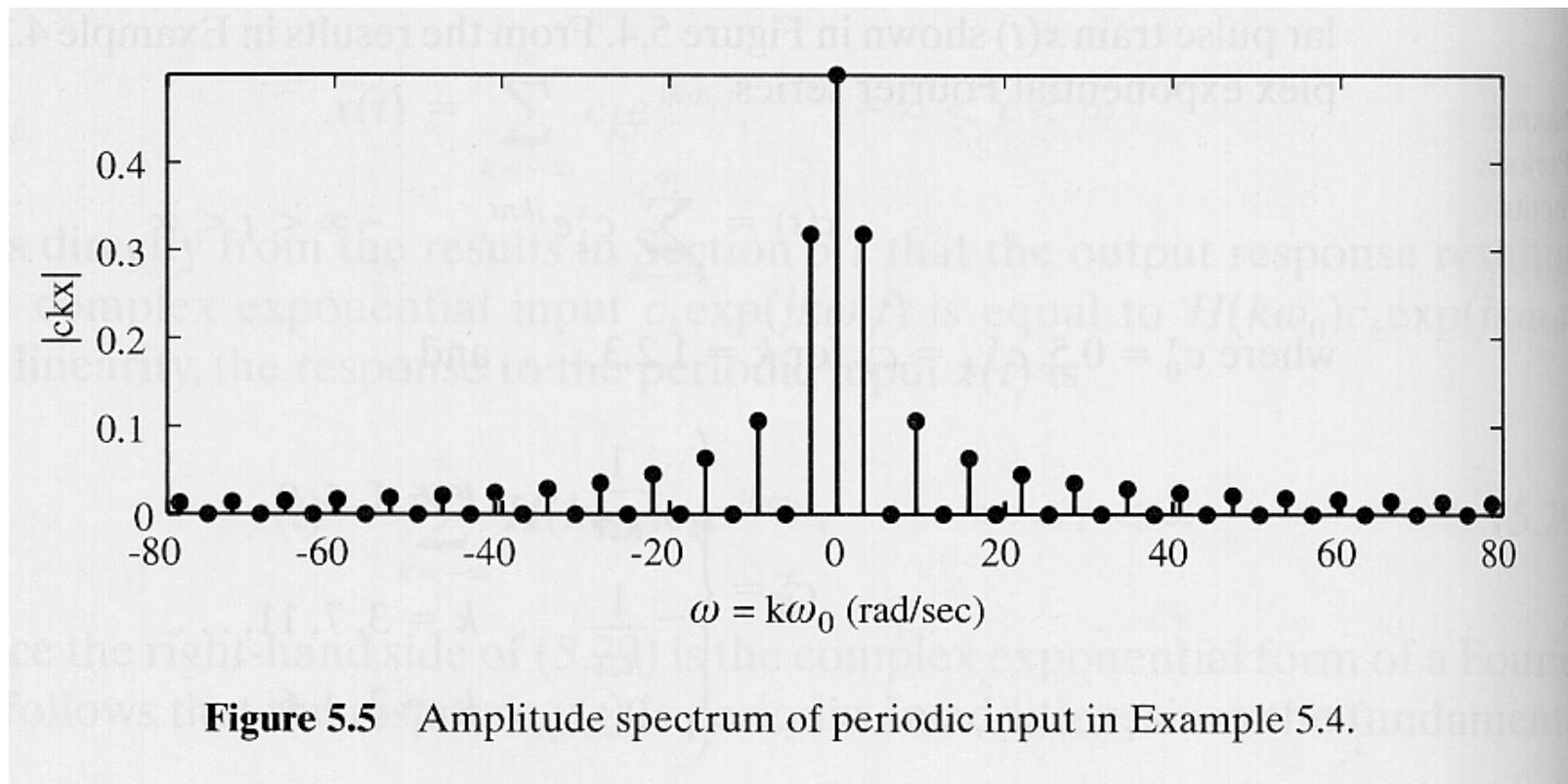
$$x(t) = \sum_{k \in \mathbb{Z}} c_k^x e^{jk\pi t}, \quad t \in \mathbb{R}$$

with

$$c_k^x = \frac{1}{2} \operatorname{sinc} \left( \frac{k}{2} \right)$$

# Example: Response of an RC Circuit to a Rectangular Pulse Train – Cont'd

- Magnitude spectrum  $|c_k^x|$  of input signal  $x(t)$



## Example: Response of an RC Circuit to a Rectangular Pulse Train – Cont'd

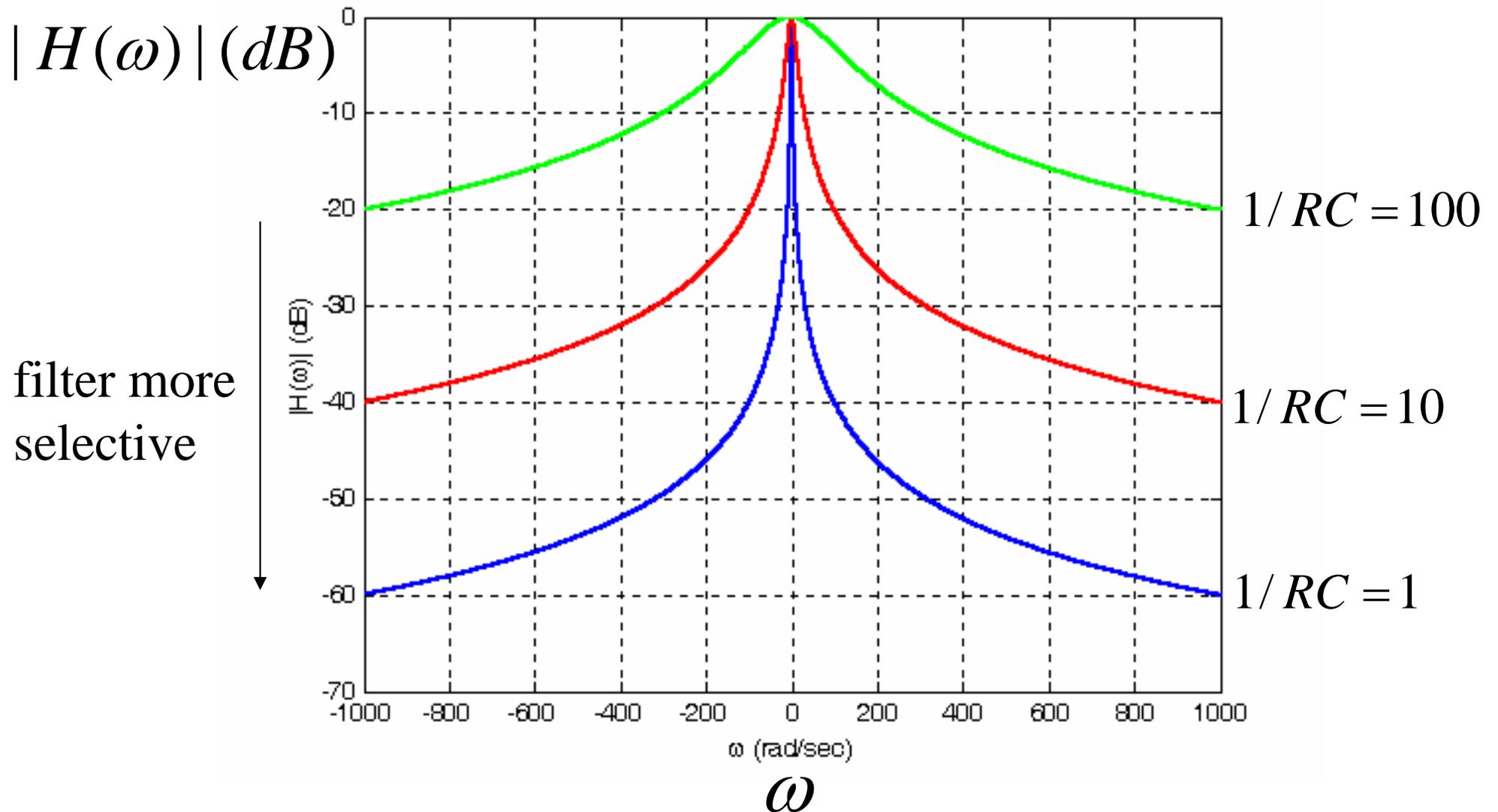
- The frequency response of the RC circuit was found to be

$$H(\omega) = \frac{1/RC}{j\omega + 1/RC}$$

- Thus, the Fourier series of the output signal is given by

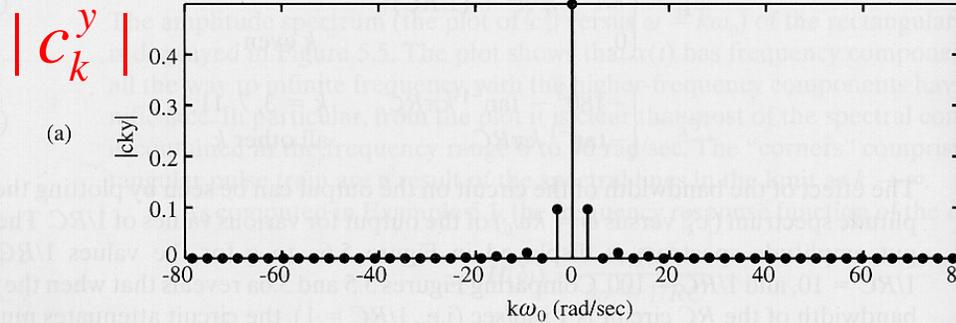
$$y(t) = \sum_{k=-\infty}^{\infty} H(k\omega_0) c_k^x e^{jk\omega_0 t} = \sum_{k=-\infty}^{\infty} c_k^y e^{jk\omega_0 t}$$

# Example: Response of an RC Circuit to a Rectangular Pulse Train – Cont'd

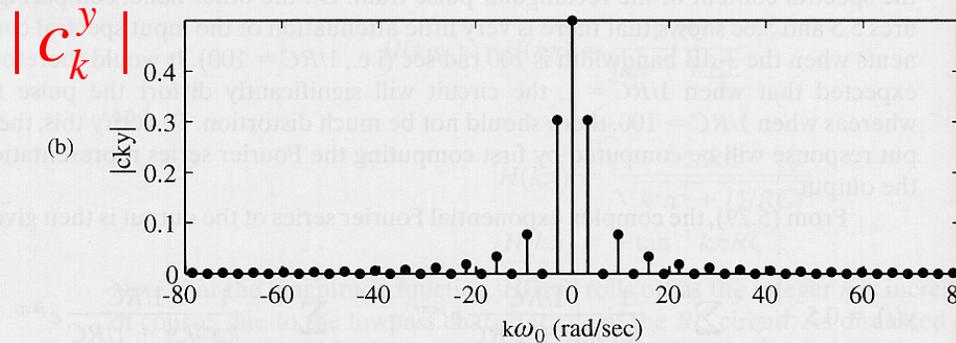


# Example: Response of an RC Circuit to a Rectangular Pulse Train – Cont'd

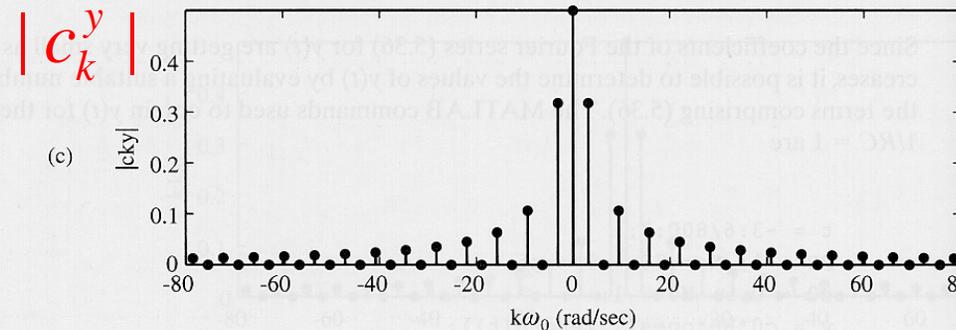
$$1/RC = 1$$



$$1/RC = 10$$



$$1/RC = 100$$

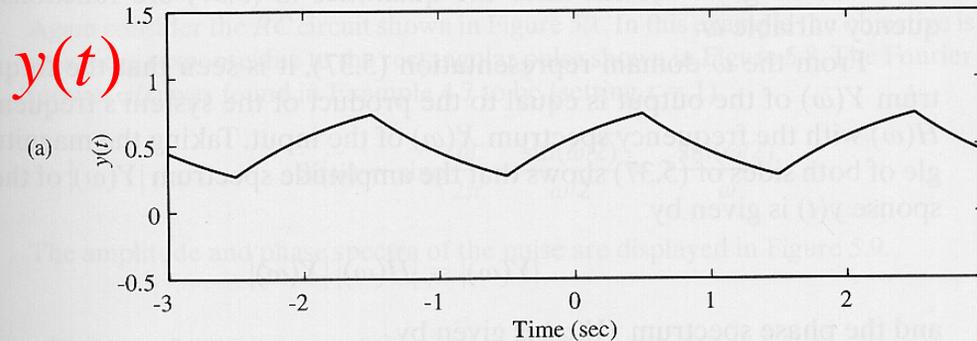


↑  
filter more selective

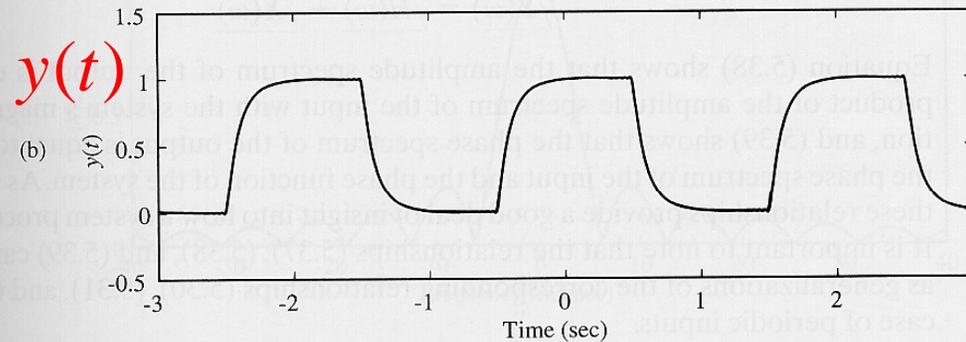
**Figure 5.6** Amplitude spectrum of output when (a)  $1/RC = 1$ ; (b)  $1/RC = 10$ ; (c)  $1/RC = 100$ .

# Example: Response of an RC Circuit to a Rectangular Pulse Train – Cont'd

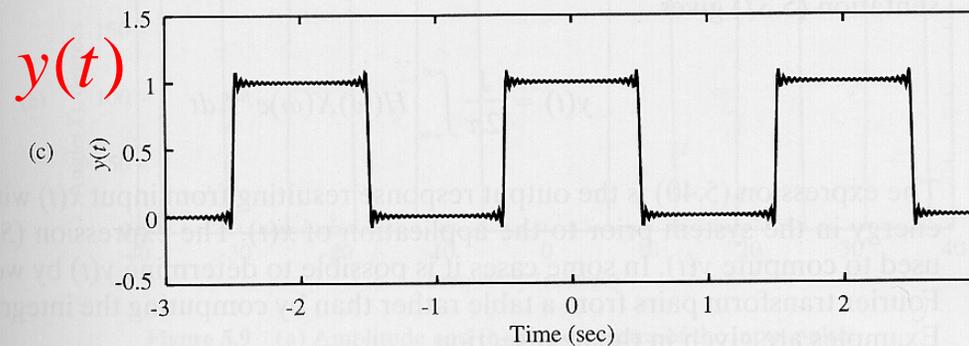
$$1/RC = 1$$



$$1/RC = 10$$



$$1/RC = 100$$

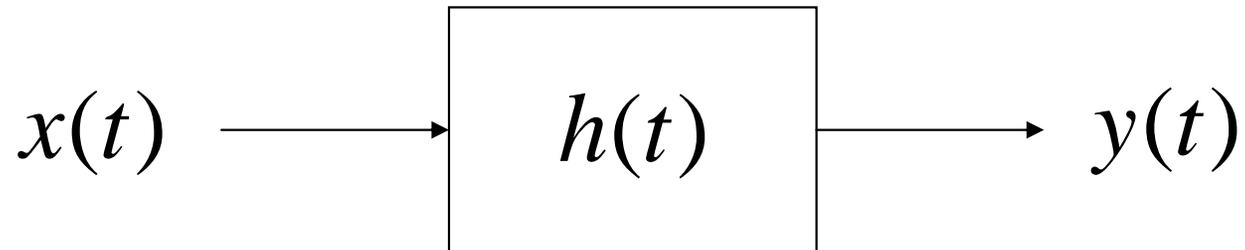


↑  
filter more selective

**Figure 5.7** Plot of output when (a)  $1/RC = 1$ ; (b)  $1/RC = 10$ ; (c)  $1/RC = 100$ .

# Response of a CT, LTI System to Aperiodic Inputs

- Consider the following CT, LTI system



- Its I/O relation is given by

$$y(t) = h(t) * x(t)$$

which, in the frequency domain, becomes

$$Y(\omega) = H(\omega)X(\omega)$$

## Response of a CT, LTI System to Aperiodic Inputs – Cont'd

- From  $Y(\omega) = H(\omega)X(\omega)$ , the **magnitude spectrum** of the output signal  $y(t)$  is given by

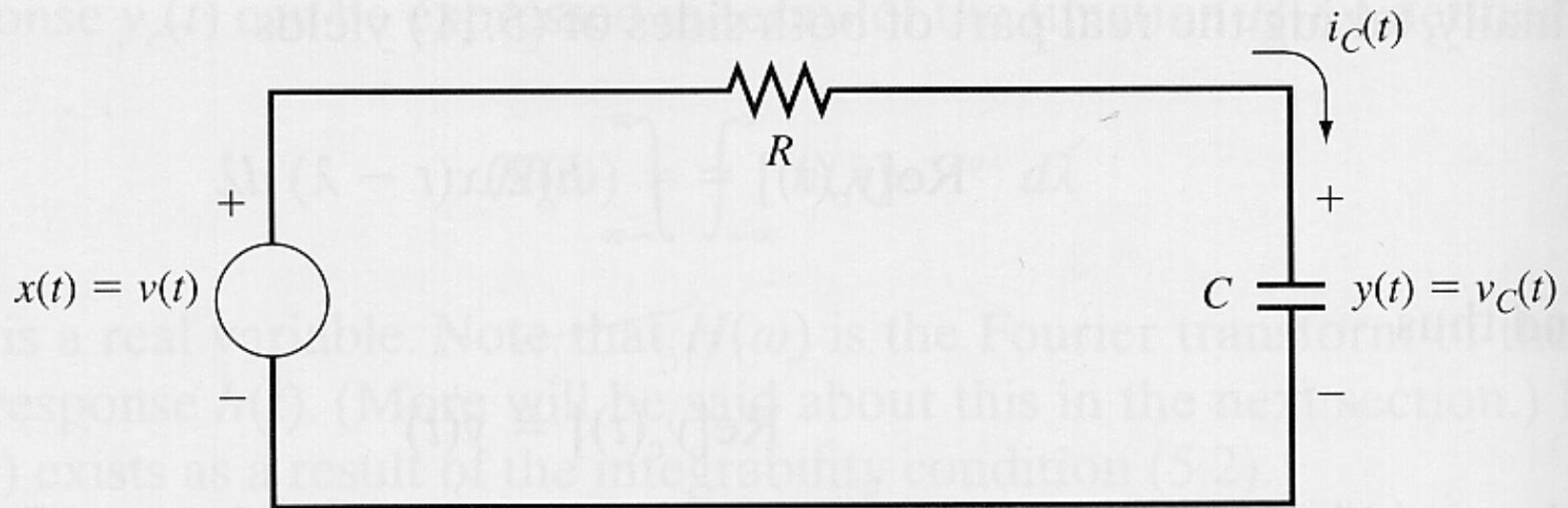
$$|Y(\omega)| = |H(\omega)| |X(\omega)|$$

and its **phase spectrum** is given by

$$\arg Y(\omega) = \arg H(\omega) + \arg X(\omega)$$

# Example: Response of an RC Circuit to a Rectangular Pulse

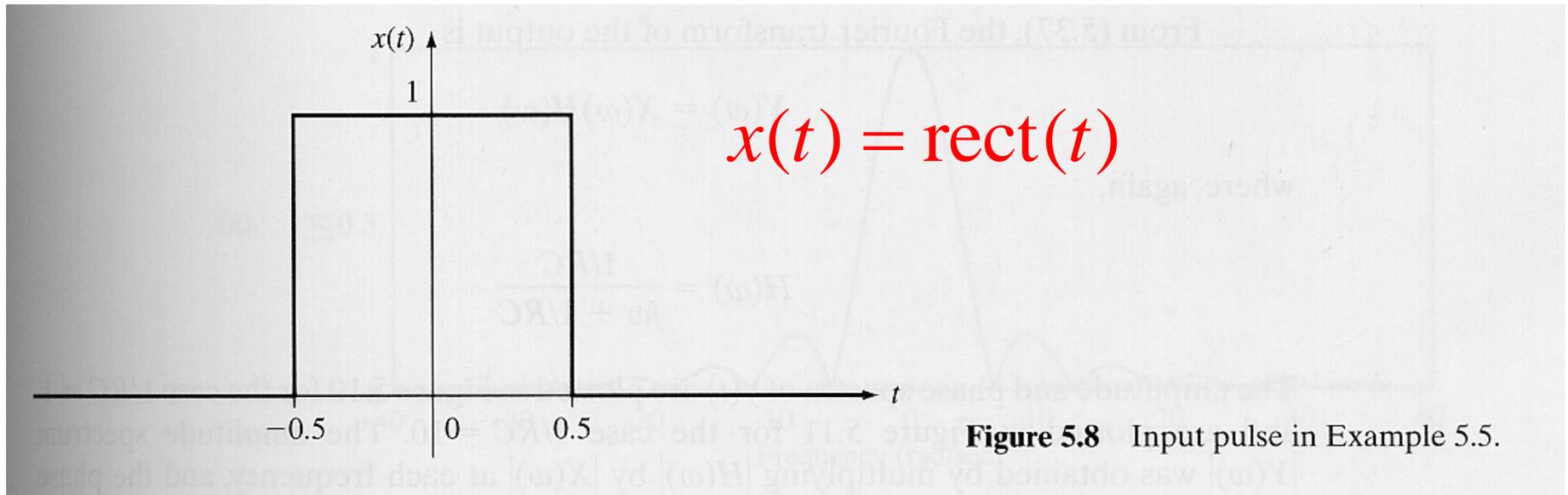
- Consider the RC circuit



**Figure 5.1** RC circuit in Example 5.2.

with input  $x(t) = \text{rect}(t)$

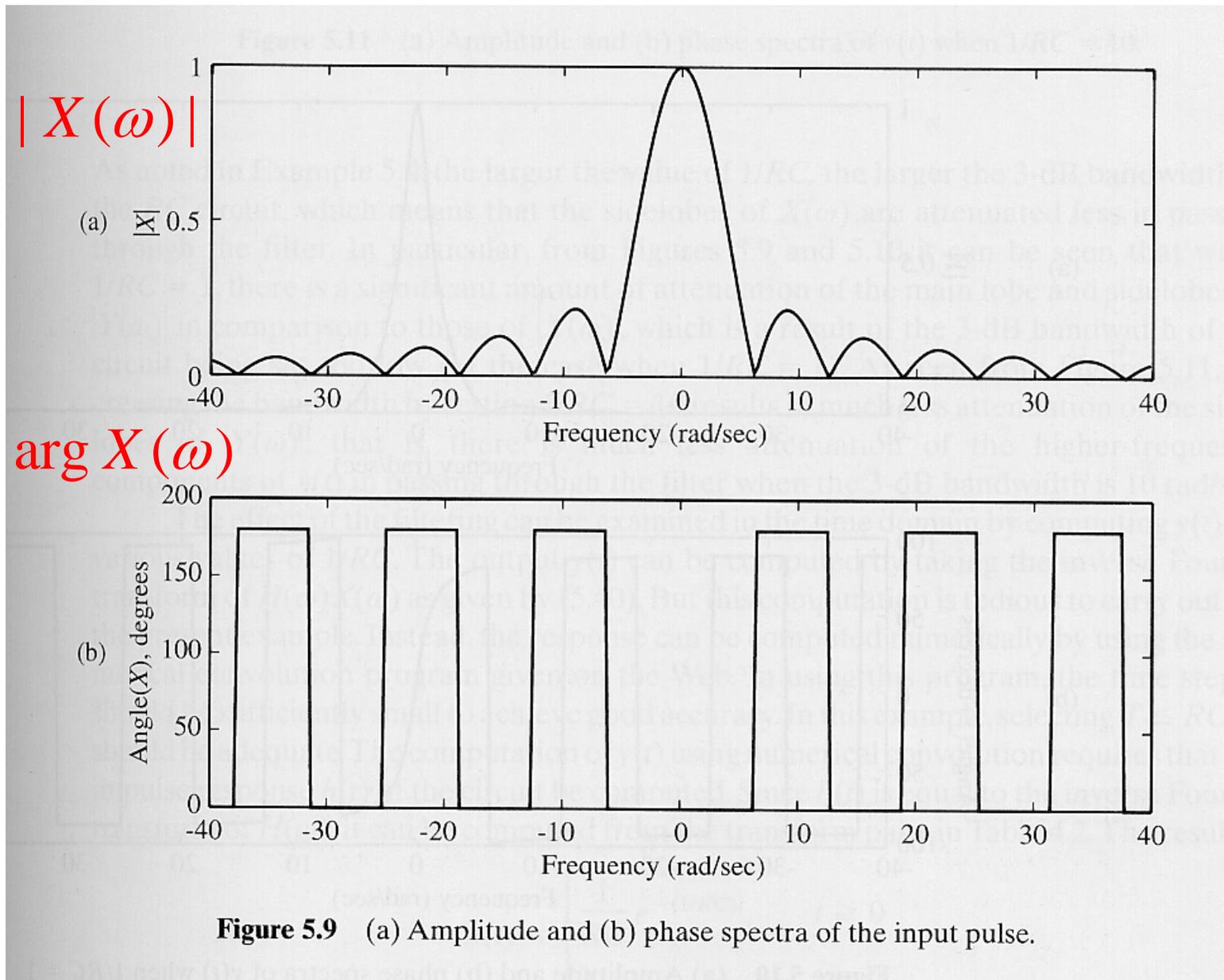
# Example: Response of an RC Circuit to a Rectangular Pulse – Cont'd



- The Fourier transform of  $x(t)$  is

$$X(\omega) = \text{sinc}\left(\frac{\omega}{2\pi}\right)$$

# Example: Response of an RC Circuit to a Rectangular Pulse – Cont'd

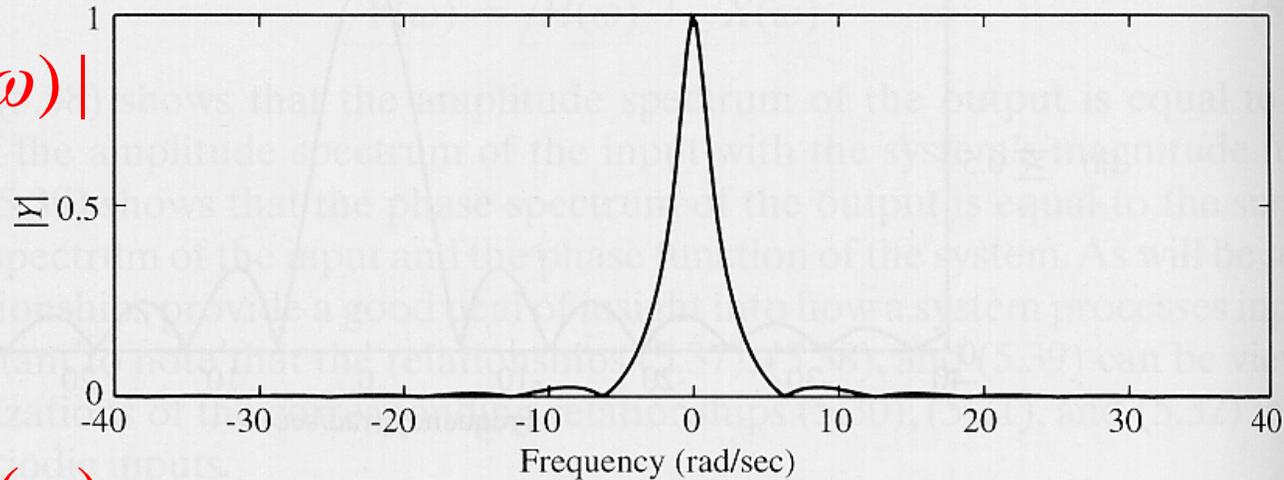


# Example: Response of an RC Circuit to a Rectangular Pulse – Cont'd

$$1/RC = 1$$

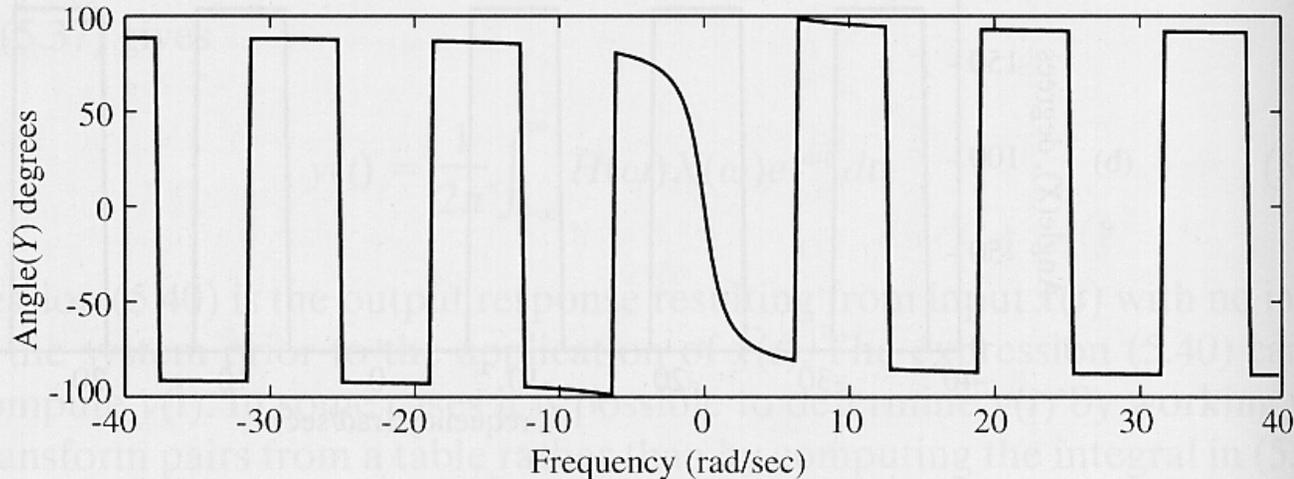
$$|Y(\omega)|$$

(a)



$$\arg Y(\omega)$$

(b)



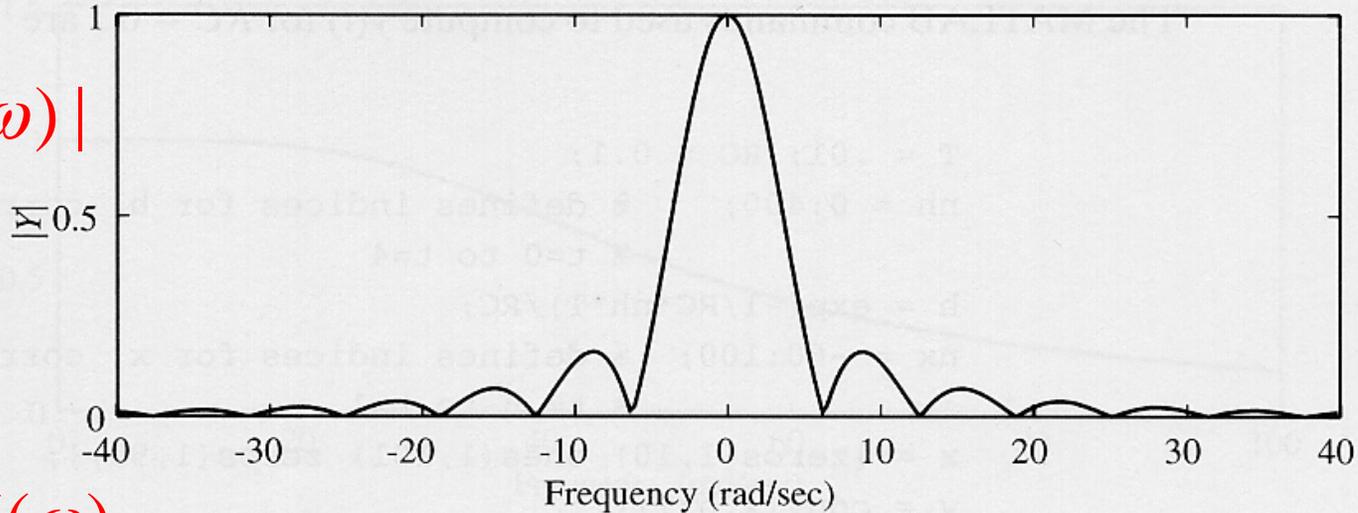
**Figure 5.10** (a) Amplitude and (b) phase spectra of  $y(t)$  when  $1/RC = 1$ .

# Example: Response of an RC Circuit to a Rectangular Pulse – Cont'd

$$1/RC = 10$$

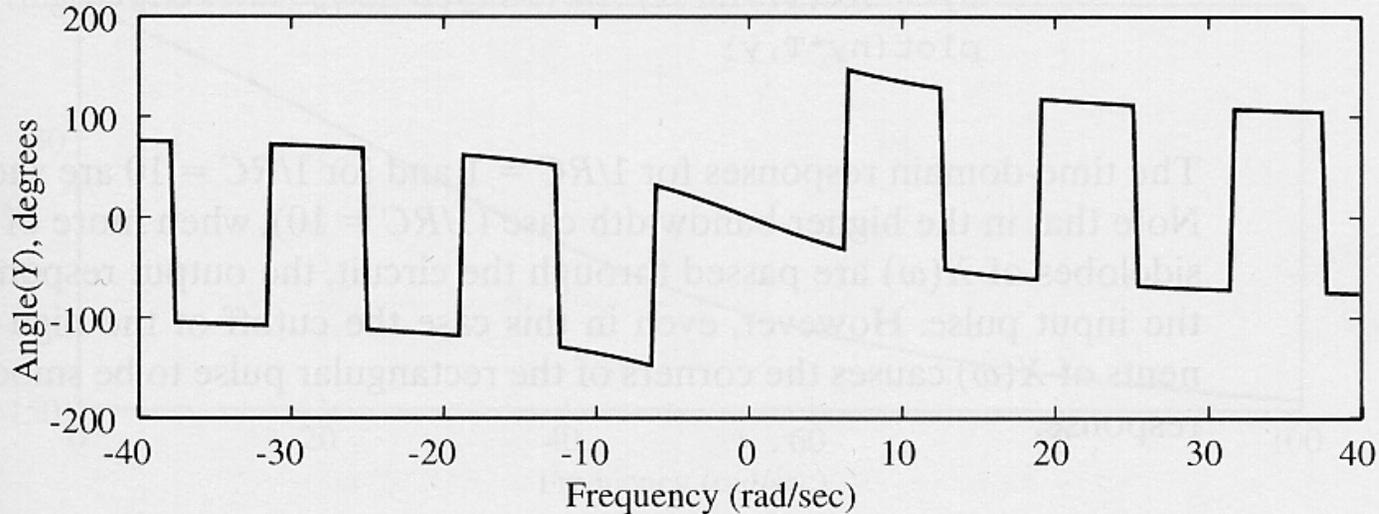
$$|Y(\omega)|$$

(a)



$$\arg Y(\omega)$$

(b)



**Figure 5.11** (a) Amplitude and (b) phase spectra of  $y(t)$  when  $1/RC = 10$ .

## Example: Response of an RC Circuit to a Rectangular Pulse – Cont'd

- The response of the system in the time domain can be found by computing the convolution

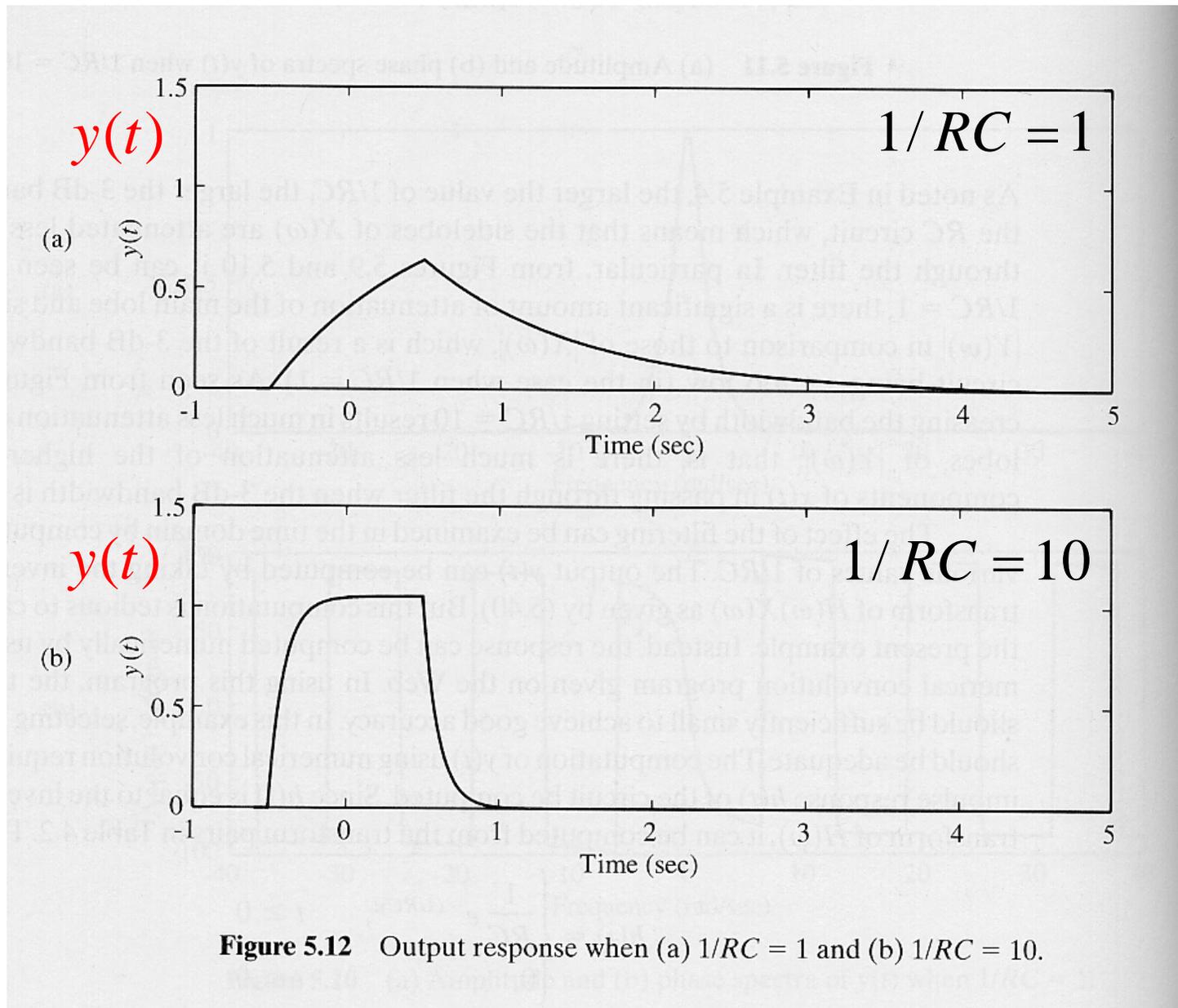
$$y(t) = h(t) * x(t)$$

where

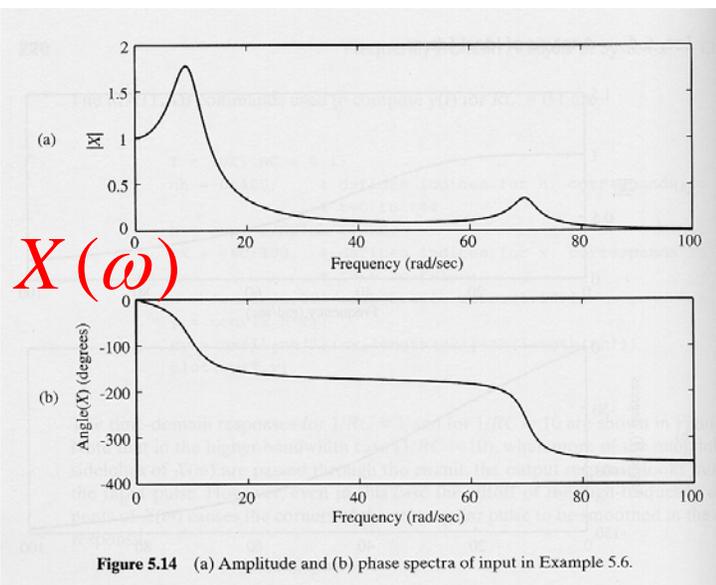
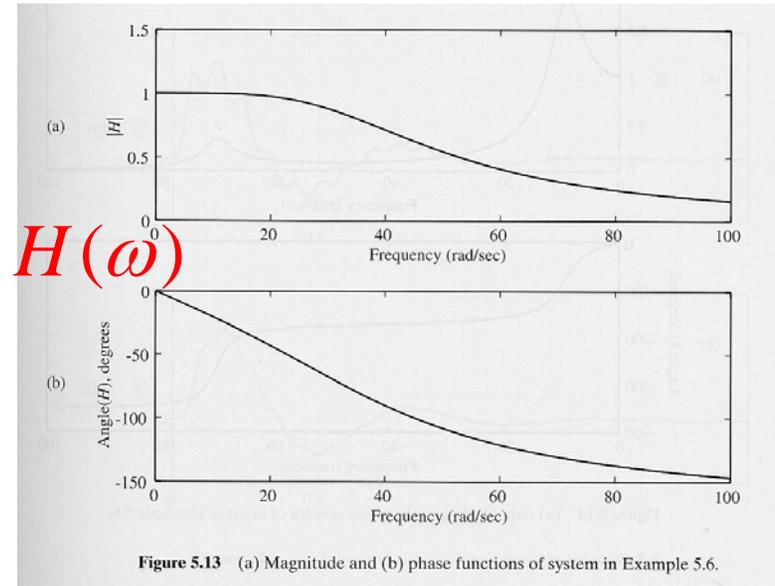
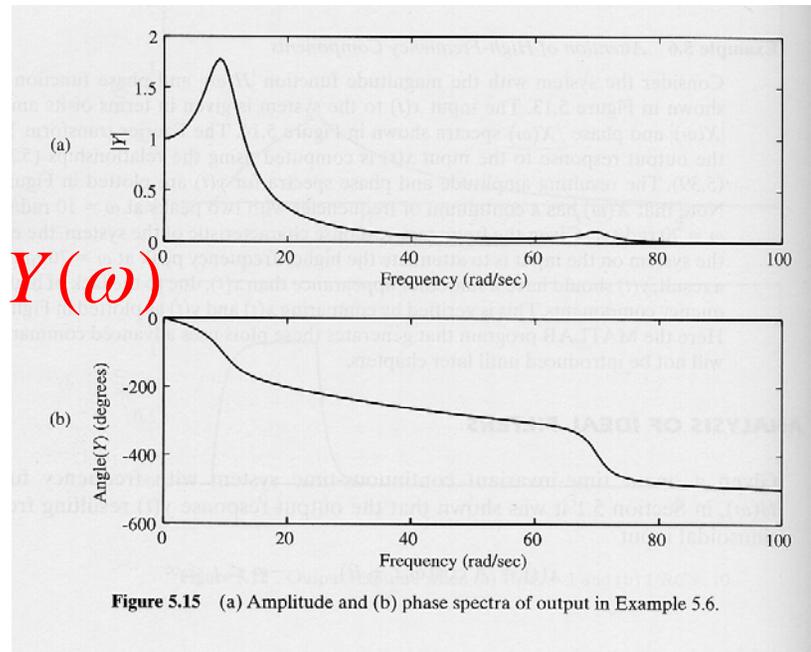
$$h(t) = (1/RC)e^{-(1/RC)t}u(t)$$

$$x(t) = \text{rect}(t)$$

# Example: Response of an RC Circuit to a Rectangular Pulse – Cont'd



# Example: Attenuation of High-Frequency Components



$Y(\omega)$

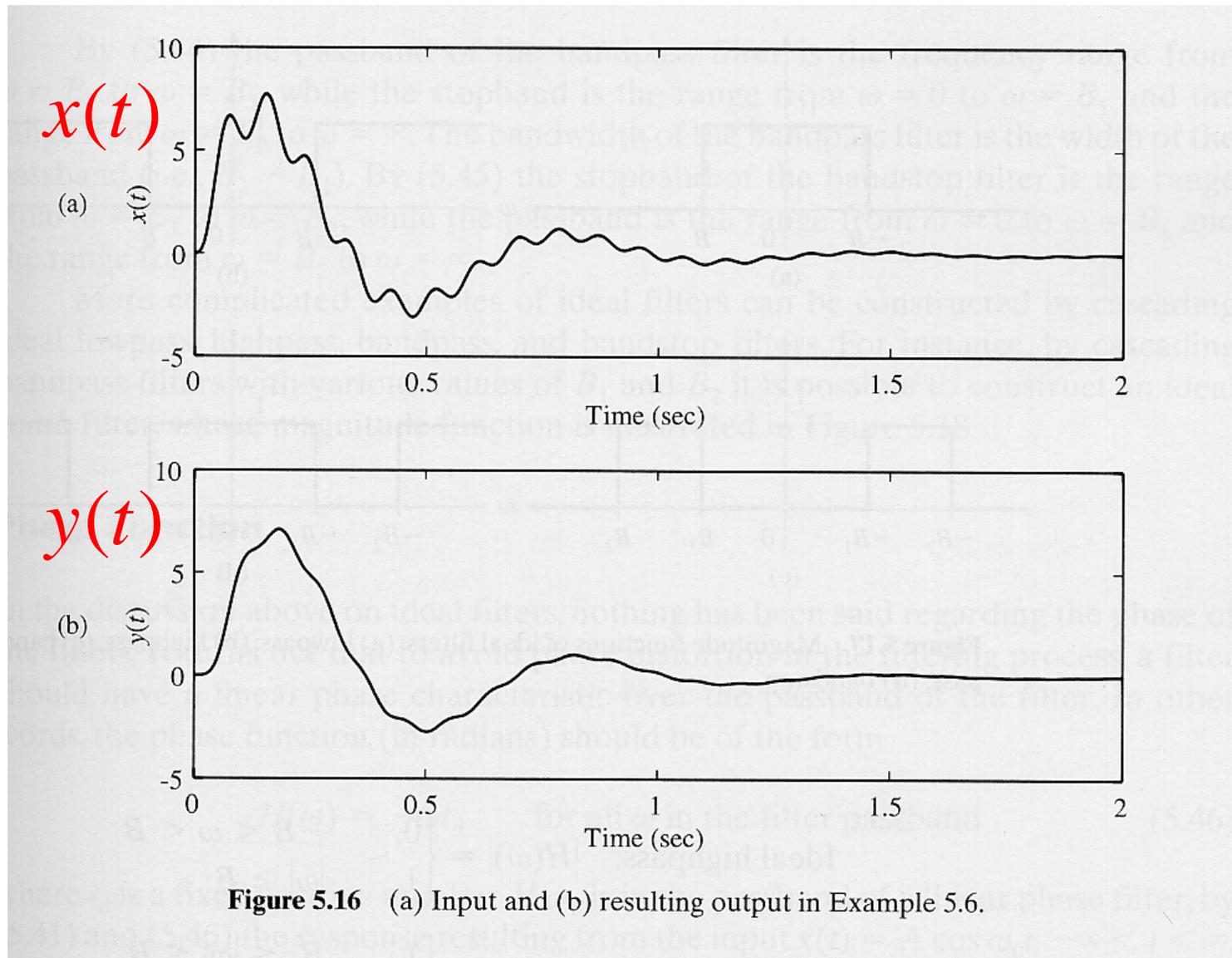
$H(\omega)$

$X(\omega)$

=

×

# Example: Attenuation of High-Frequency Components



# Filtering Signals

- The response of a CT, LTI system with frequency response  $H(\omega)$  to a sinusoidal signal

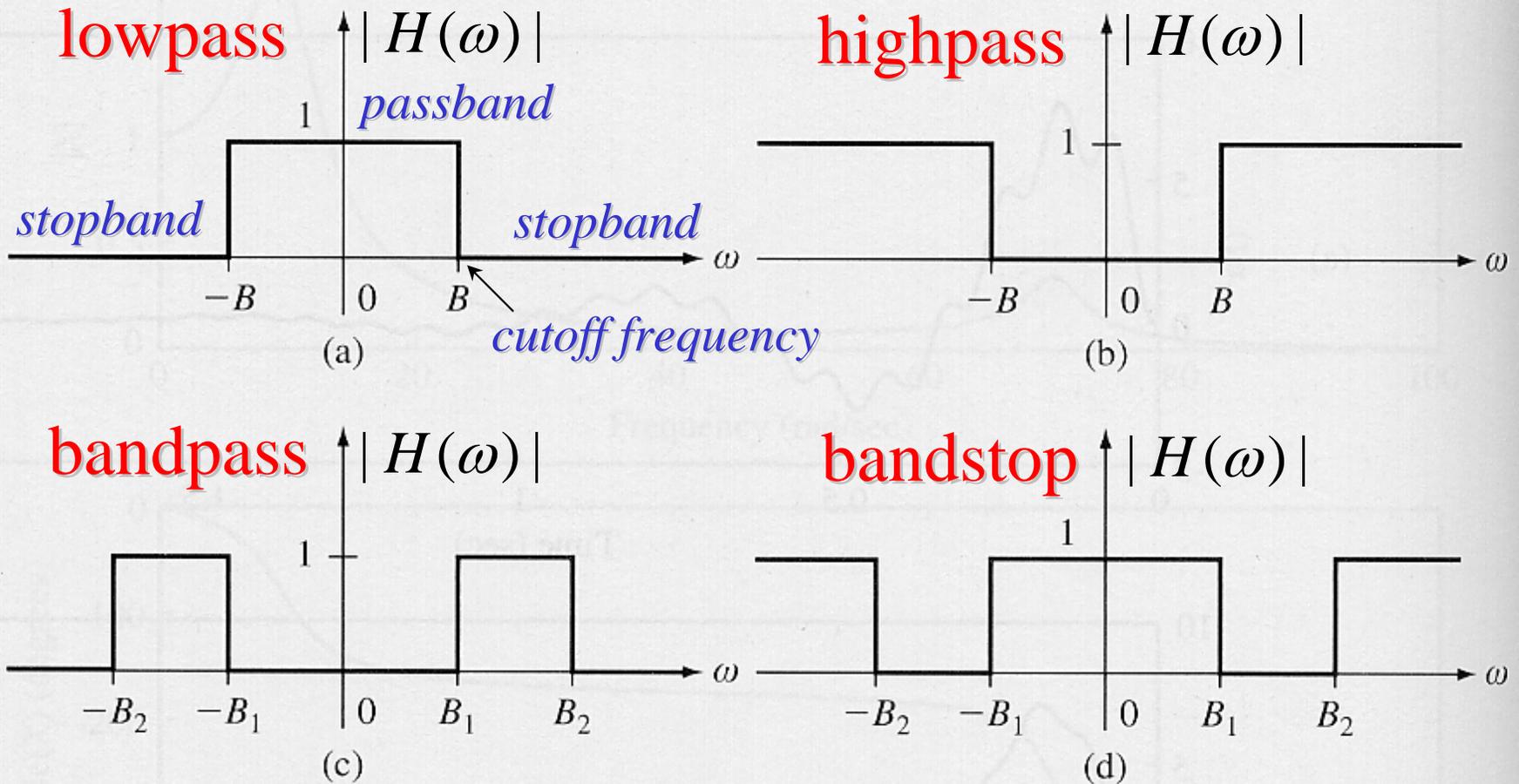
$$x(t) = A \cos(\omega_0 t + \theta)$$

is

$$y(t) = A |H(\omega_0)| \cos(\omega_0 t + \theta + \arg H(\omega_0))$$

- **Filtering**: if  $|H(\omega_0)| = 0$  or  $|H(\omega_0)| \approx 0$  then  $y(t) = 0$  or  $y(t) \approx 0, \forall t \in \mathbb{R}$

# Four Basic Types of Filters



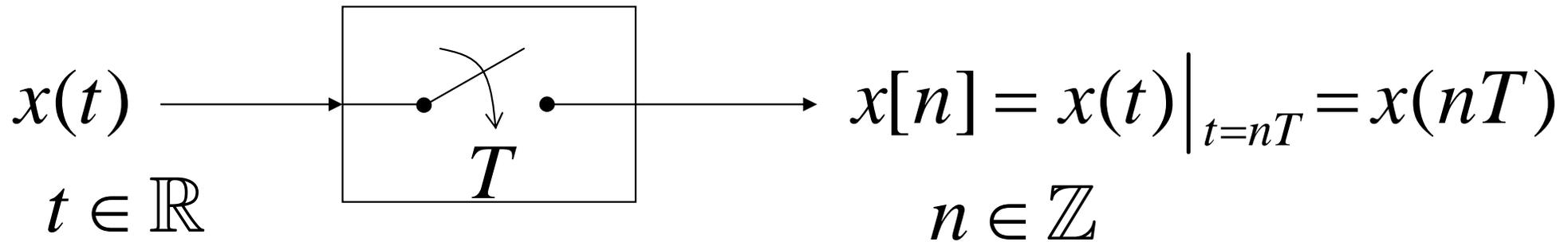
**Figure 5.17** Magnitude functions of ideal filters: (a) lowpass; (b) highpass; (c) bandpass; (d) bandstop.

# Phase Function

- Filters are usually designed based on specifications on the magnitude response  $|H(\omega)|$
- The phase response  $\arg H(\omega)$  has to be taken into account too in order to prevent signal distortion as the signal goes through the system
- If the filter has **linear phase** in its passband(s), then there is **no distortion**

# Ideal Sampling

- Consider the ideal sampler:



- It is convenient to express the sampled signal  $x(nT)$  as  $x(t)p(t)$  where

$$p(t) = \sum_{n \in \mathbb{Z}} \delta(t - nT)$$

## Ideal Sampling – Cont'd

- Thus, the sampled waveform  $x(t)p(t)$  is

$$x(t)p(t) = \sum_{n \in \mathbb{Z}} x(t)\delta(t - nT) = \sum_{n \in \mathbb{Z}} x(nT)\delta(t - nT)$$

- $x(t)p(t)$  is an impulse train whose weights (areas) are the sample values  $x(nT)$  of the original signal  $x(t)$

## Ideal Sampling – Cont'd

- Since  $p(t)$  is periodic with period  $T$ , it can be represented by its **Fourier series**

$$p(t) = \sum_{k \in \mathbb{Z}} c_k e^{jk\omega_s t}, \quad \omega_s = \frac{2\pi}{T} \quad \begin{array}{l} \text{sampling} \\ \text{frequency} \\ \text{(rad/sec)} \end{array}$$

$$\begin{aligned} \text{where } c_k &= \frac{1}{T} \int_{-T/2}^{T/2} p(t) e^{-jk\omega_s t} dt, \quad k \in \mathbb{Z} \\ &= \frac{1}{T} \int_{-T/2}^{T/2} \delta(t) e^{-jk\omega_s t} dt = \frac{1}{T} \end{aligned}$$

# Ideal Sampling – Cont'd

- Therefore

$$p(t) = \sum_{k \in \mathbb{Z}} \frac{1}{T} e^{jk\omega_s t}$$

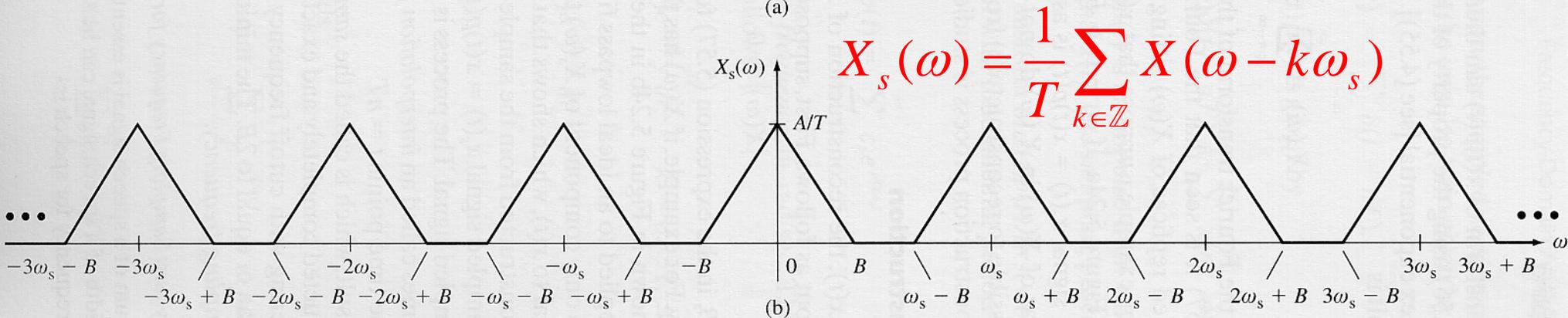
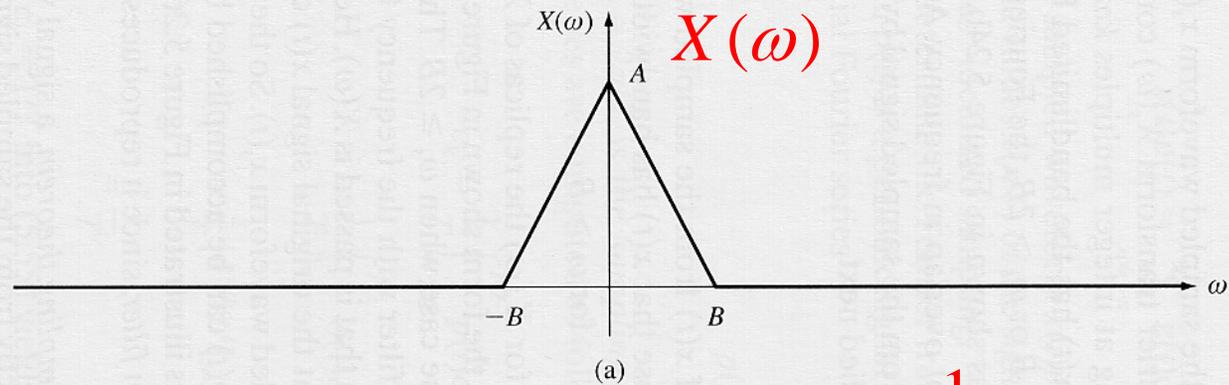
and

$$x_s(t) = x(t)p(t) = \sum_{k \in \mathbb{Z}} \frac{1}{T} x(t) e^{jk\omega_s t} = \frac{1}{T} \sum_{k \in \mathbb{Z}} x(t) e^{jk\omega_s t}$$

whose Fourier transform is

$$X_s(\omega) = \frac{1}{T} \sum_{k \in \mathbb{Z}} X(\omega - k\omega_s)$$

# Ideal Sampling – Cont'd



**Figure 5.24** Fourier transform of (a)  $x(t)$  and (b)  $x_s(t) = x(t)p(t)$ .

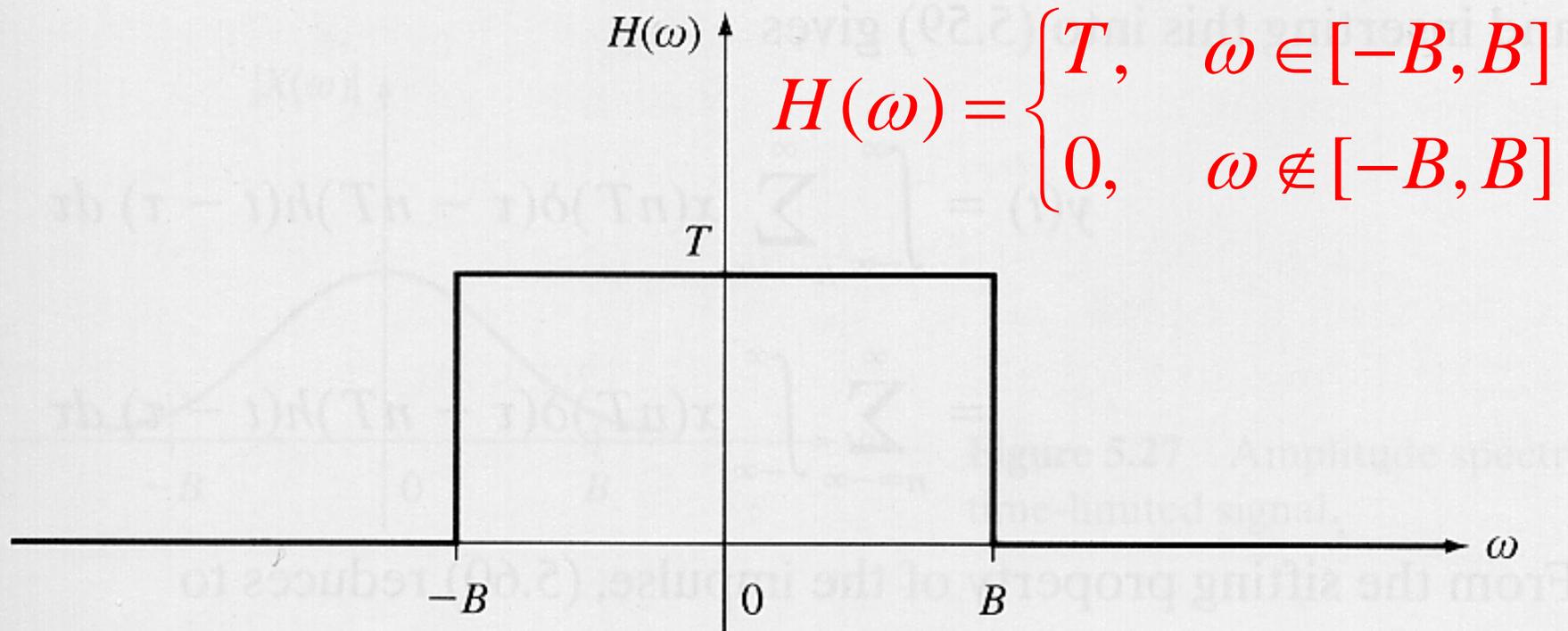
# Signal Reconstruction

- Suppose that the signal  $x(t)$  is bandlimited with bandwidth  $B$ , i.e.,  $|X(\omega)| = 0$ , for  $|\omega| > B$
- Then, if  $\omega_s \geq 2B$ , the replicas of  $X(\omega)$  in

$$X_s(\omega) = \frac{1}{T} \sum_{k \in \mathbb{Z}} X(\omega - k\omega_s)$$

do not overlap and  $X(\omega)$  can be recovered by applying an ideal lowpass filter to  $X_s(\omega)$   
(**interpolation filter**)

# Interpolation Filter for Signal Reconstruction



**Figure 5.25** Frequency response function of ideal lowpass filter with bandwidth  $B$ .

# Interpolation Formula

- The impulse response  $h(t)$  of the interpolation filter is

$$h(t) = \frac{BT}{\pi} \operatorname{sinc}\left(\frac{B}{\pi}t\right)$$

and the output  $y(t)$  of the interpolation filter is given by

$$y(t) = h(t) * x_s(t)$$

# Interpolation Formula – Cont'd

- But

$$x_s(t) = x(t)p(t) = \sum_{n \in \mathbb{Z}} x(nT)\delta(t - nT)$$

whence

$$y(t) = h(t) * x_s(t) = \sum_{n \in \mathbb{Z}} x(nT)h(t - nT) =$$

$$= \frac{BT}{\pi} \sum_{n \in \mathbb{Z}} x(nT) \operatorname{sinc} \left( \frac{B}{\pi} (t - nT) \right)$$

- Moreover,

$$y(t) = x(t)$$

# Shannon's Sampling Theorem

- A CT bandlimited signal  $x(t)$  with frequencies no higher than  $B$  can be reconstructed from its samples  $x[n] = x(nT)$  if the samples are taken at a rate

$$\omega_s = 2\pi / T \geq 2B$$

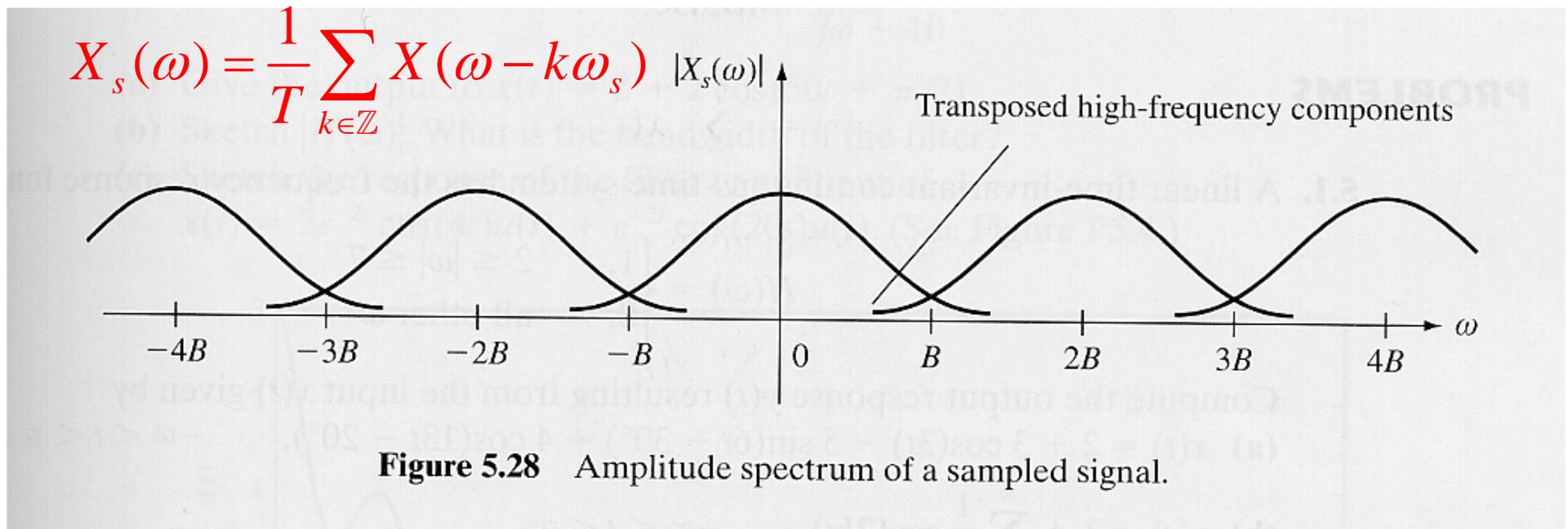
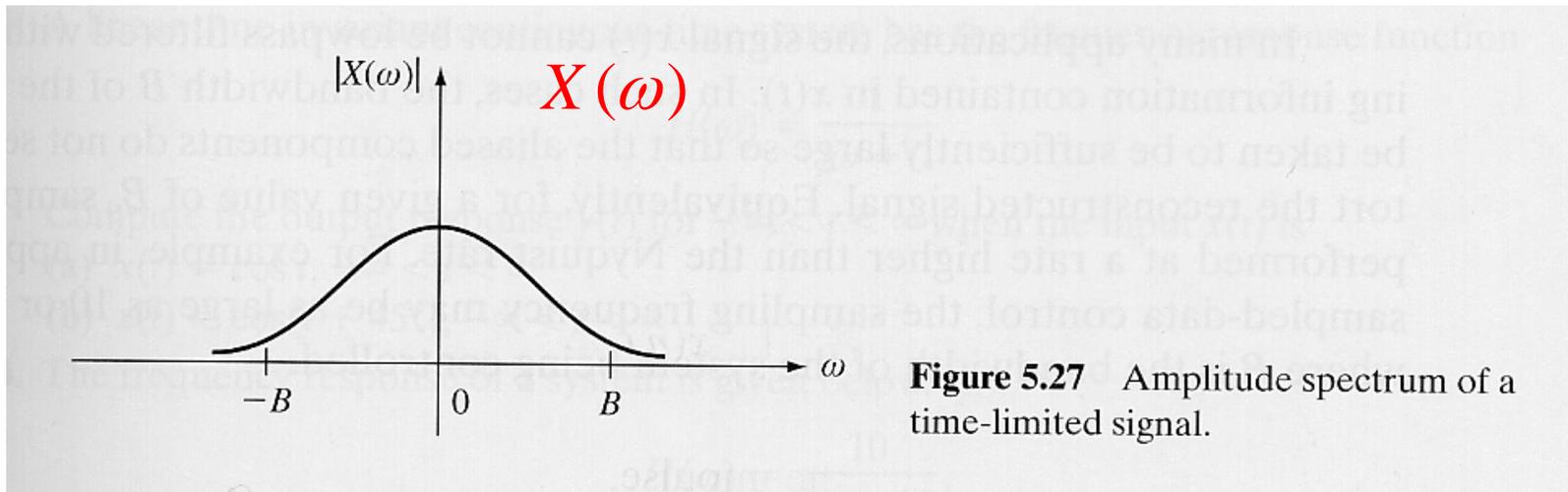
- The reconstruction of  $x(t)$  from its samples  $x[n] = x(nT)$  is provided by the interpolation formula

$$x(t) = \frac{BT}{\pi} \sum_{n \in \mathbb{Z}} x(nT) \operatorname{sinc} \left( \frac{B}{\pi} (t - nT) \right)$$

# Nyquist Rate

- The minimum sampling rate  $\omega_s = 2\pi / T = 2B$  is called the Nyquist rate
- Question: Why do CD's adopt a sampling rate of 44.1 *kHz*?
- Answer: Since the highest frequency perceived by humans is about 20 *kHz*, 44.1 *kHz* is slightly more than twice this upper bound

# Aliasing



## Aliasing –Cont'd

- Because of aliasing, it is not possible to reconstruct  $x(t)$  exactly by lowpass filtering the sampled signal  $x_s(t) = x(t)p(t)$
- Aliasing results in a distorted version of the original signal  $x(t)$
- It can be eliminated (theoretically) by lowpass filtering  $x(t)$  before sampling it so that  $|X(\omega)| = 0$  for  $|\omega| \geq B$