



ANALOG SIGNAL PROCESSING



ECE 210 & 211
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Objectives

- **LTI System Response to Energy Signals**
- **Amplitude modulation**
- **Coherent demodulation of AM signal**

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- **LTI System Response to Energy Signals**
- **Amplitude modulation**
- **Coherent demodulation of AM signal**

LTI Circuit Response

- Previously, we obtained the input – Output relation

$$f(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} F(\omega) e^{j\omega t} d\omega$$

LTI

$$y(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} H(\omega) F(\omega) e^{j\omega t} d\omega$$

LTI Circuit Response

- Alternatively, we can write this relation as,

$$y(t) \longleftrightarrow Y(\omega) = H(\omega)F(\omega)$$

- The system output is the IFT (Inverse Fourier Transform) of the product of the system frequency response and Fourier transform of the input
- The relation is simply drawn from,



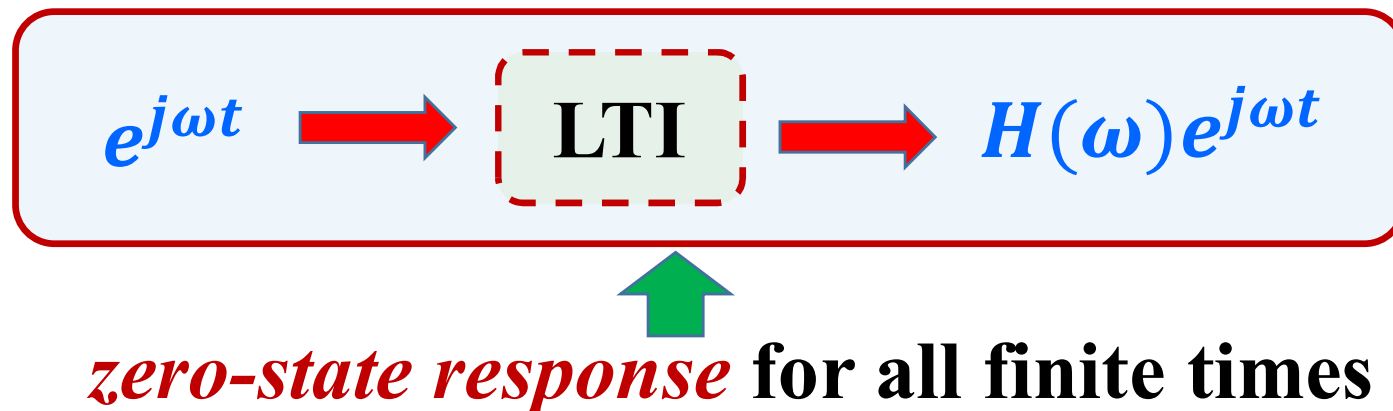
LTI Circuit Response

$$y(t) \longleftrightarrow Y(\omega) = H(\omega)F(\omega)$$

- The relation describes just steady state response $y(t)$ to a dissipative LTI system response to an input

$$f(t) \longleftrightarrow F(\omega)$$

- In dissipative systems, transient part of zero-state response to input $\cos(\omega t)$ and $\sin(\omega t)$ applied at $t = -\infty$ must be vanished for finite times



LTI Circuit Response

- The Inverse Fourier Transform of the

$$y(t) \longleftrightarrow Y(\omega) = H(\omega)F(\omega)$$

Represents , for all finite t, *the entire* (**steady state + transient**) zero state response of the system $H(\omega)$ to the input

$$f(t) \longleftrightarrow F(\omega)$$

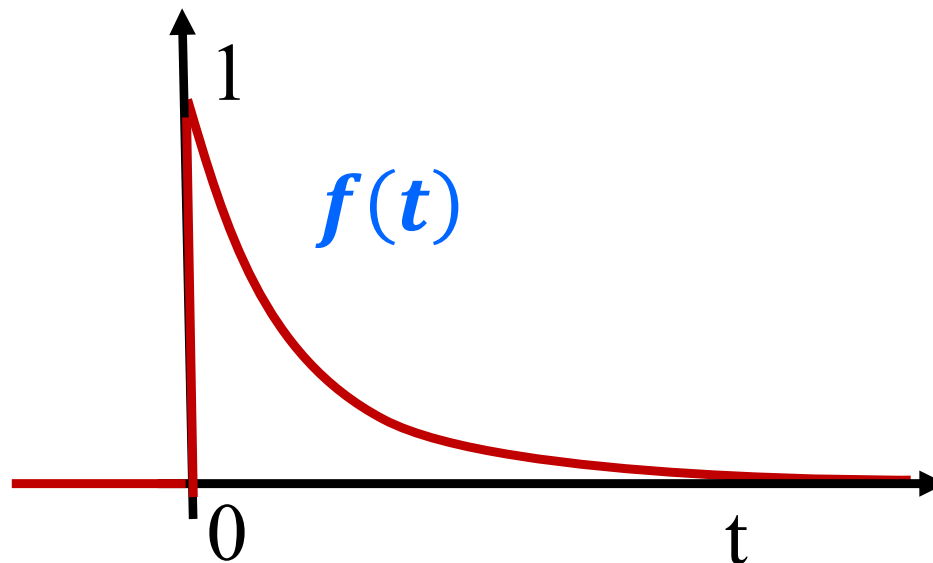
Dissipative LTI Response – Example 1

Question: The input of an LTI system,

is

$$H(\omega) = \frac{1}{1 + j\omega}$$
$$f(t) = e^{-t}u(t)$$

Determine the *zero-state response* $y(t)$?



Dissipative LTI Response – Example 1

Solution: Since,

$$f(t) = e^{-t}u(t) \longleftrightarrow F(\omega) = \frac{1}{1 + j\omega}$$

The Fourier transform of $y(t)$,

$$Y(\omega) = H(\omega)F(\omega) = \frac{1}{1 + j\omega} \frac{1}{1 + j\omega} = \frac{1}{(1 + j\omega)^2}$$

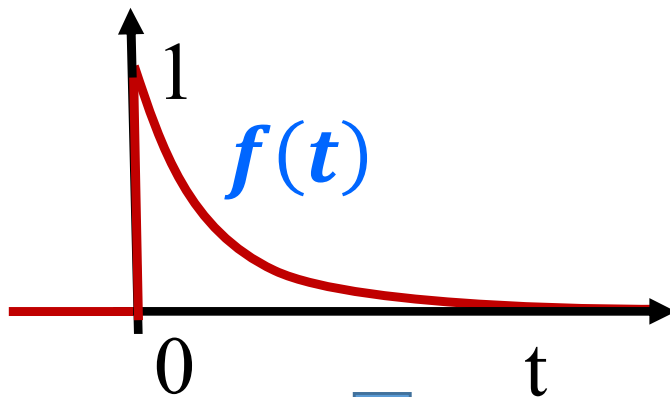
From Transform pairs,

$$t \cdot e^{-t}u(t) \longleftrightarrow \frac{1}{(1 + j\omega)^2}$$

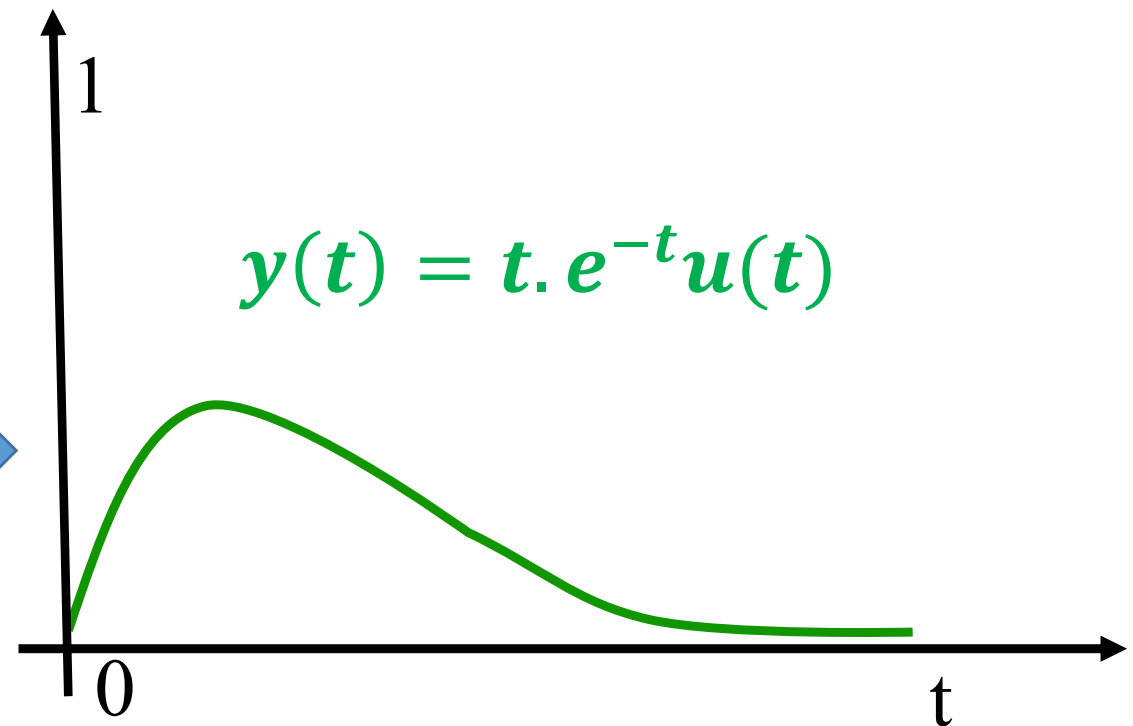
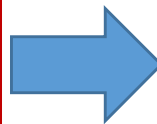
Thus,

$$y(t) = t \cdot e^{-t}u(t)$$

Dissipative LTI Response – Example 1



**LTI zero-state
response**



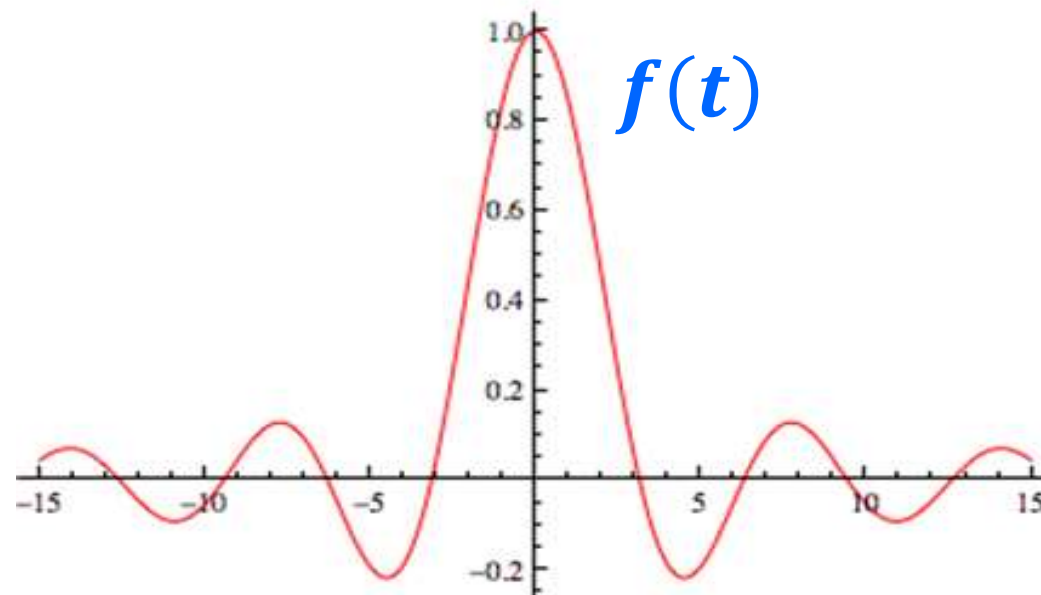
Dissipative LTI Response – Example 2

Question: The input of an LTI system,

$$f(t) = \text{sinc}(t) \longleftrightarrow F(\omega) = \pi \text{rect}\left(\frac{\omega}{2}\right)$$

having the frequency response, $H(\omega) = \text{rect}(\omega)$

Determine the *zero-state response* $y(t)$?



Dissipative LTI Response – Example 2

Solution: Given that,

$$F(\omega) = \pi \text{rect}\left(\frac{\omega}{2}\right) \quad \text{and} \quad H(\omega) = \text{rect}(\omega)$$

We have,
$$Y(\omega) = H(\omega)F(\omega) = \text{rect}(\omega) \cdot \pi \text{rect}\left(\frac{\omega}{2}\right) \\ = \pi \text{rect}(\omega)$$

taking the inverse Fourier transform of $Y(\omega)$,

$$y(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} Y(\omega) e^{j\omega t} d\omega = \frac{1}{2\pi} \int_{-\infty}^{\infty} \pi \text{rect}(\omega) e^{j\omega t} d\omega$$

Dissipative LTI Response – Example 2

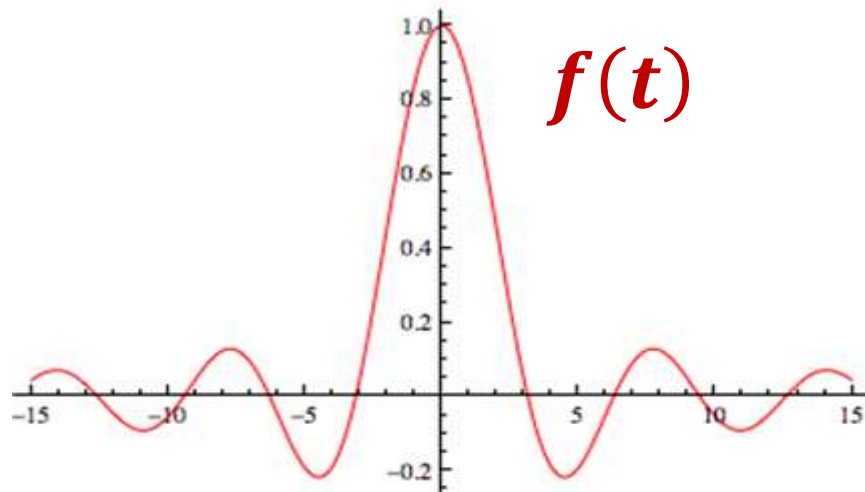
$$y(t) = \frac{1}{2} \int_{-\frac{1}{2}}^{\frac{1}{2}} e^{j\omega t} d\omega = \frac{e^{j\frac{t}{2}} - e^{-j\frac{t}{2}}}{2jt}$$

the result simplifies to,

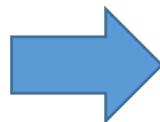
$$y(t) = \frac{1}{2} \text{sinc} \left(\frac{t}{2} \right)$$

- The system broadens the input $f(t)$ by a factor of 2 by having its bandwidth

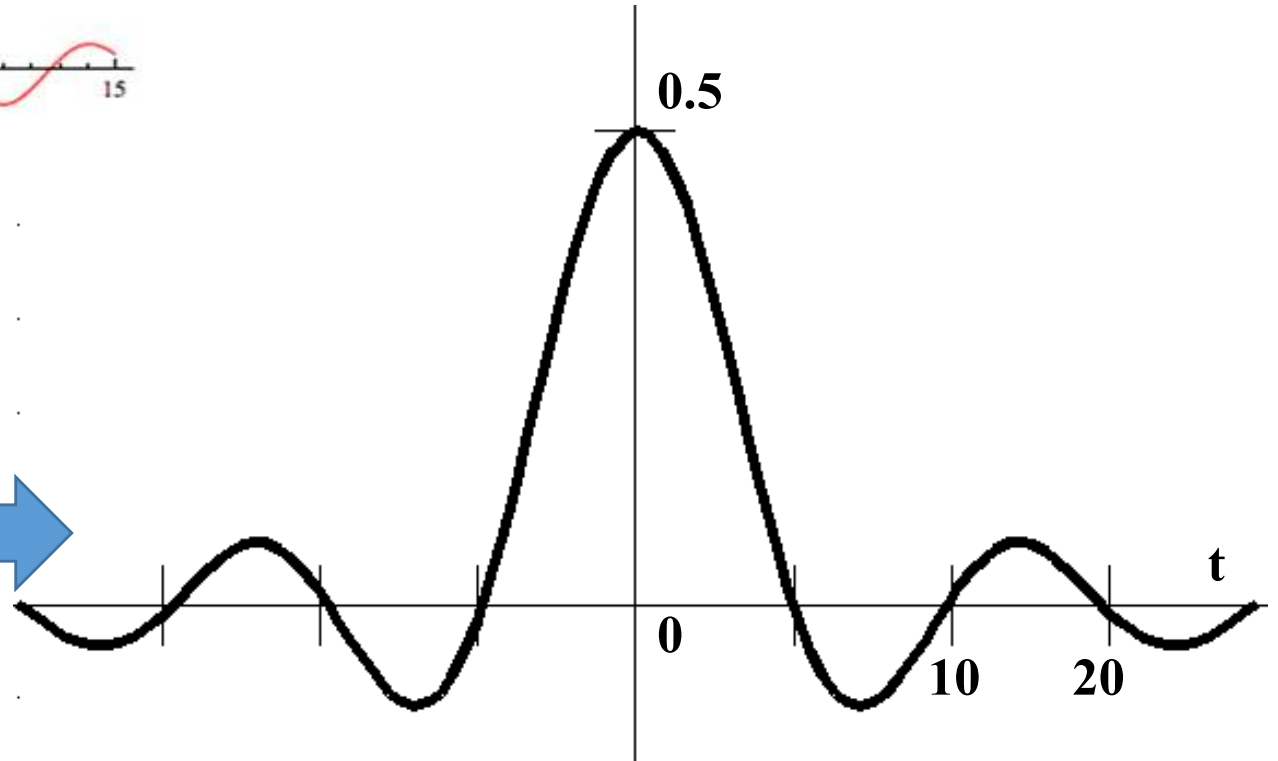
Dissipative LTI Response – Example 2



**LTI zero-state
response**



$$y(t) = \frac{1}{2} \text{sinc} \left(\frac{t}{2} \right)$$



Dissipative LTI Response – Example 3

Question: The signal

$$f(t) = g_1(t) + g_2(t)$$

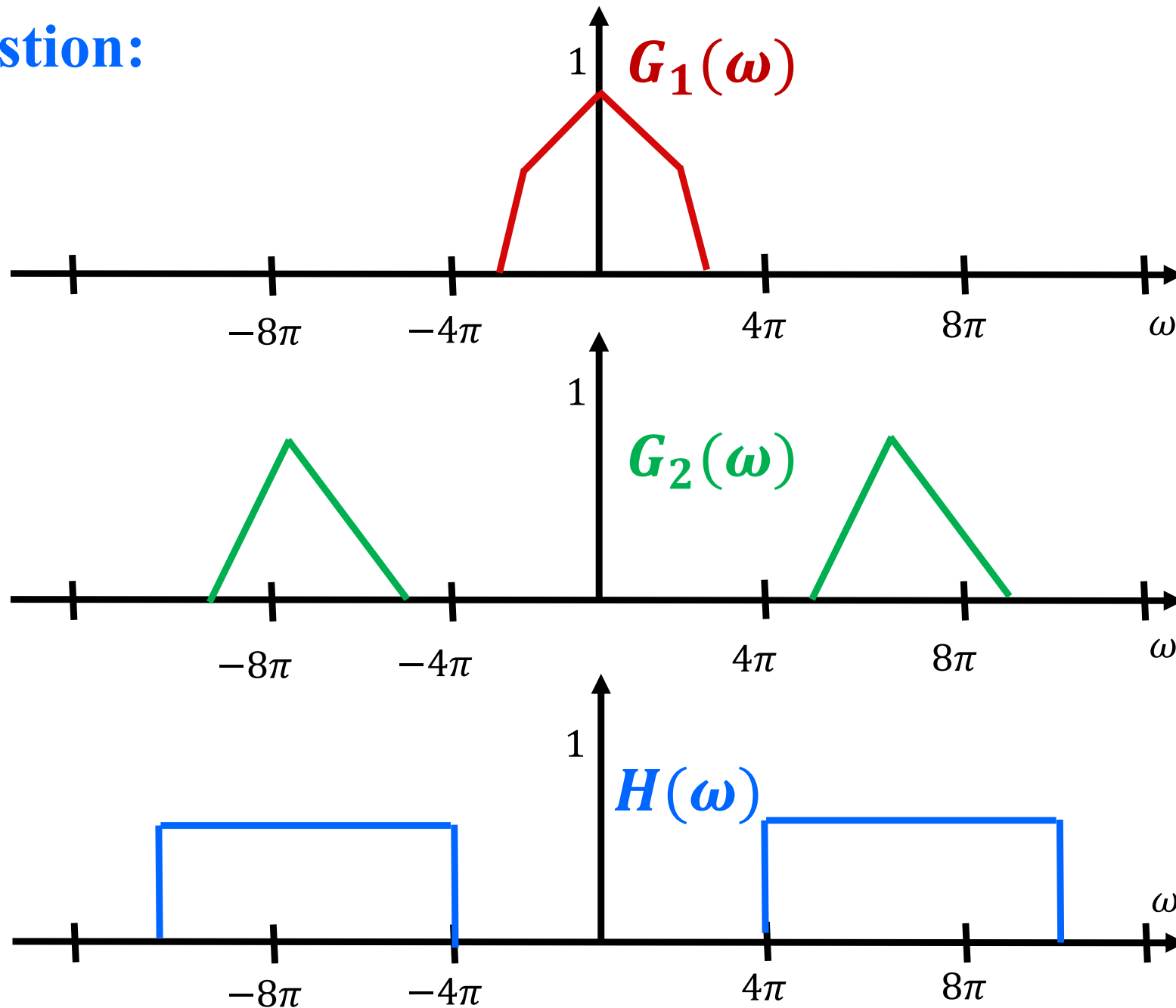
is passed through a bandpass filter having frequency response $H(\omega)$ shown.

Determine the zero-state response $y(t)$ in terms of $g_1(t)$ and $g_2(t)$?

$G_1(\omega)$ and $G_2(\omega)$ are Fourier transform of $g_1(t)$ and $g_2(t)$, respectively.

Dissipative LTI Response – Example 3

Question:



Dissipative LTI Response – Example 3

Solution: Since,

$$f(t) = g_1(t) + g_2(t)$$

$$f(t) \longleftrightarrow F(\omega) = G_1(\omega) + G_2(\omega)$$

Therefore, the Fourier transform of the output $y(t)$

$$Y(\omega) = H(\omega)F(\omega) = H(\omega)G_1(\omega) + H(\omega)G_2(\omega)$$

from the figure, we can observe that,

$$H(\omega)G_1(\omega) = 0$$

Dissipative LTI Response – Example 3

and,

$$H(\omega)G_2(\omega) = \frac{1}{2}G_2(\omega)$$

Hence,

$$Y(\omega) = H(\omega)F(\omega) = \frac{1}{2}G_2(\omega)$$

So, the inverse Fourier transform will be,

$$y(t) = \frac{1}{2}g_2(t)$$

The system filters out the component $g_1(t)$ from the input and delivers a scaled-down replica of $g_2(t)$ as the output

Dissipative LTI Response – Example 4

Question: The input $f(t)$ is passed through a system having frequency response of

$$H(\omega) = e^{-jt_o\omega}$$

Determine the zero-state response $y(t)$?

Solution:

$$Y(\omega) = H(\omega)F(\omega) = F(\omega)e^{-jt_o\omega}$$

Using the time sifting property of the Fourier transforms, we found that output $y(t)$ is,

$$y(t) = f(t - t_o)$$

which is a delayed copy of the input $f(t)$

Dissipative LTI Response – Example 5

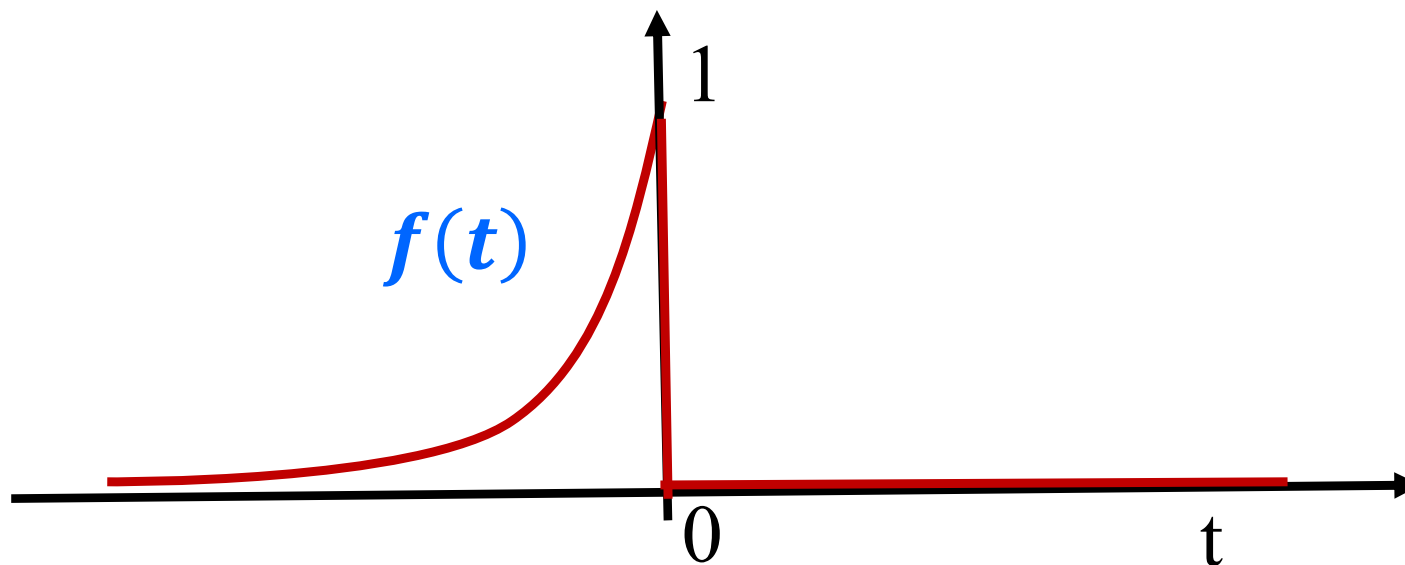
Question: The input of an LTI system,

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

is

$$f(t) = e^t u(-t)$$

Determine the *zero-state response* $y(t)$?



Dissipative LTI Response – Example 5

Solution: Since,

$$f(t) = e^{-t}u(-t) \longleftrightarrow F(\omega) = \frac{1}{1 - j\omega}$$

The Fourier transform of $y(t)$,

$$Y(\omega) = H(\omega)F(\omega) = \frac{1}{1 + j\omega} \frac{1}{1 - j\omega} = \frac{1}{1 + \omega^2}$$

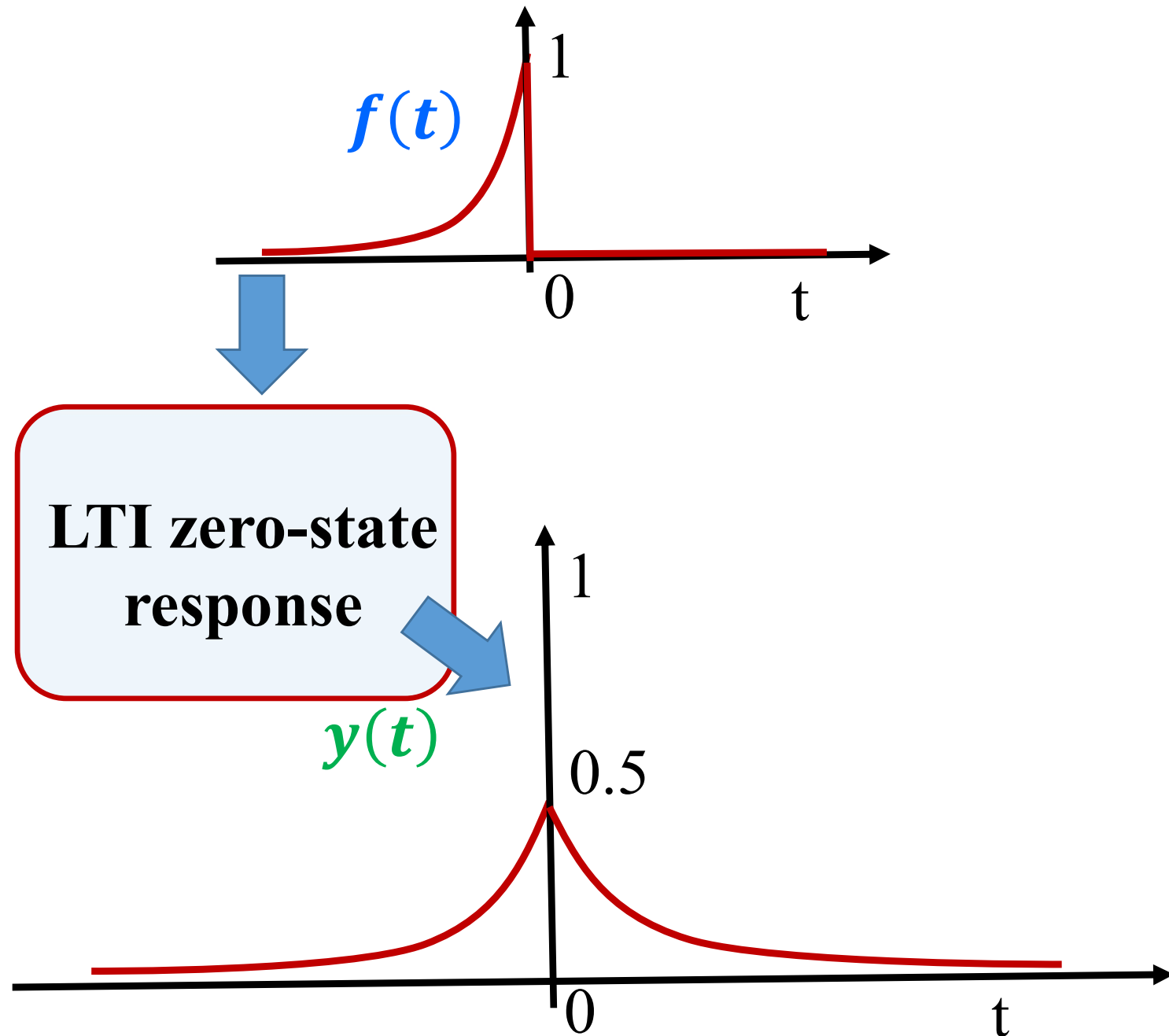
From Transform pairs,

$$e^{-|t|} \longleftrightarrow \frac{2}{(1 + j\omega)^2}$$

Thus output will be,

$$y(t) = \frac{1}{2} e^{-|t|}$$

Dissipative LTI Response – Example 5



Dissipative LTI Response – Example 6

Question: The input of an LTI system,

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

is

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

What is the system output $y(t)$?

Dissipative LTI Response – Example 6

Solution: Since,

$$f(t) = \text{rect}(t) \longleftrightarrow F(\omega) = \text{sinc}\left(\frac{\omega}{2}\right)$$

The Fourier transform of $y(t)$,

$$Y(\omega) = H(\omega)F(\omega) = \frac{1}{1 + j\omega} \text{sinc}\left(\frac{\omega}{2}\right)$$

taking the inverse Fourier transform,

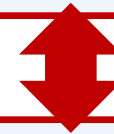
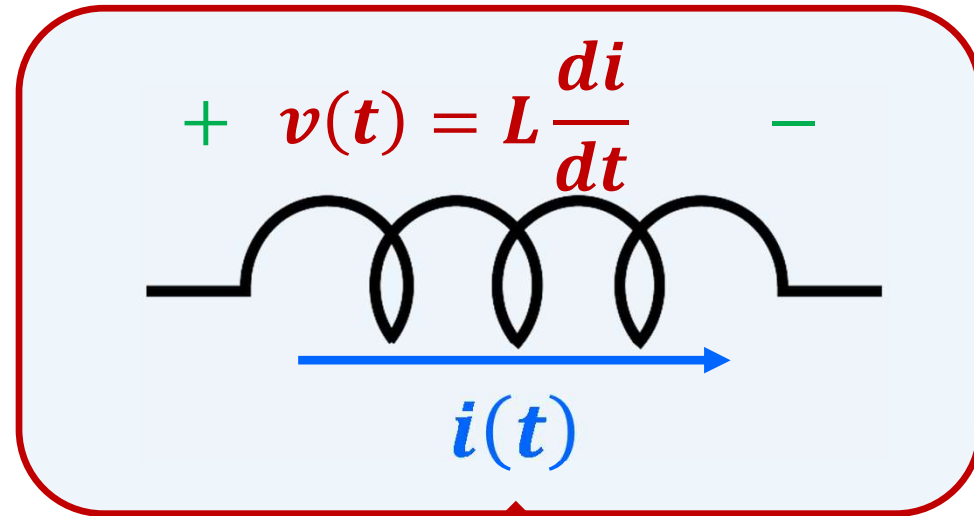
$$y(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} \frac{1}{1 + j\omega} \text{sinc}\left(\frac{\omega}{2}\right) e^{j\omega t} d\omega$$

Dissipative LTI Response for Circuits

We know that for inductor, *i-v* relation looks like,

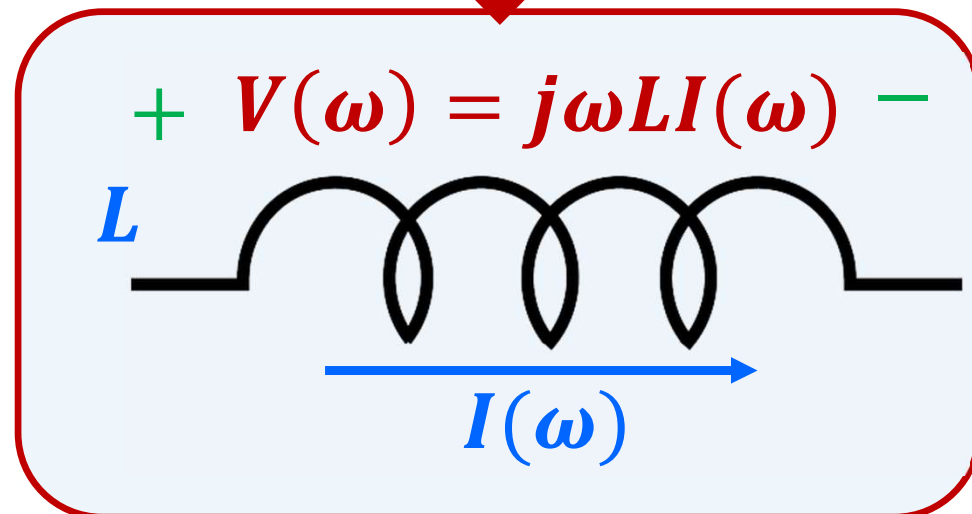
Time Domain

$$v(t) = L \frac{di}{dt}$$



Frequency Domain

$$V(\omega) = j\omega LI(\omega)$$



Dissipative LTI Response for circuits

From the correspondence $\frac{di}{dt} \longleftrightarrow j\omega I(\omega)$

Amplitude-scaling property implies that,

$$V(\omega) = j\omega L I(\omega) \Rightarrow V(\omega) = Z I(\omega)$$

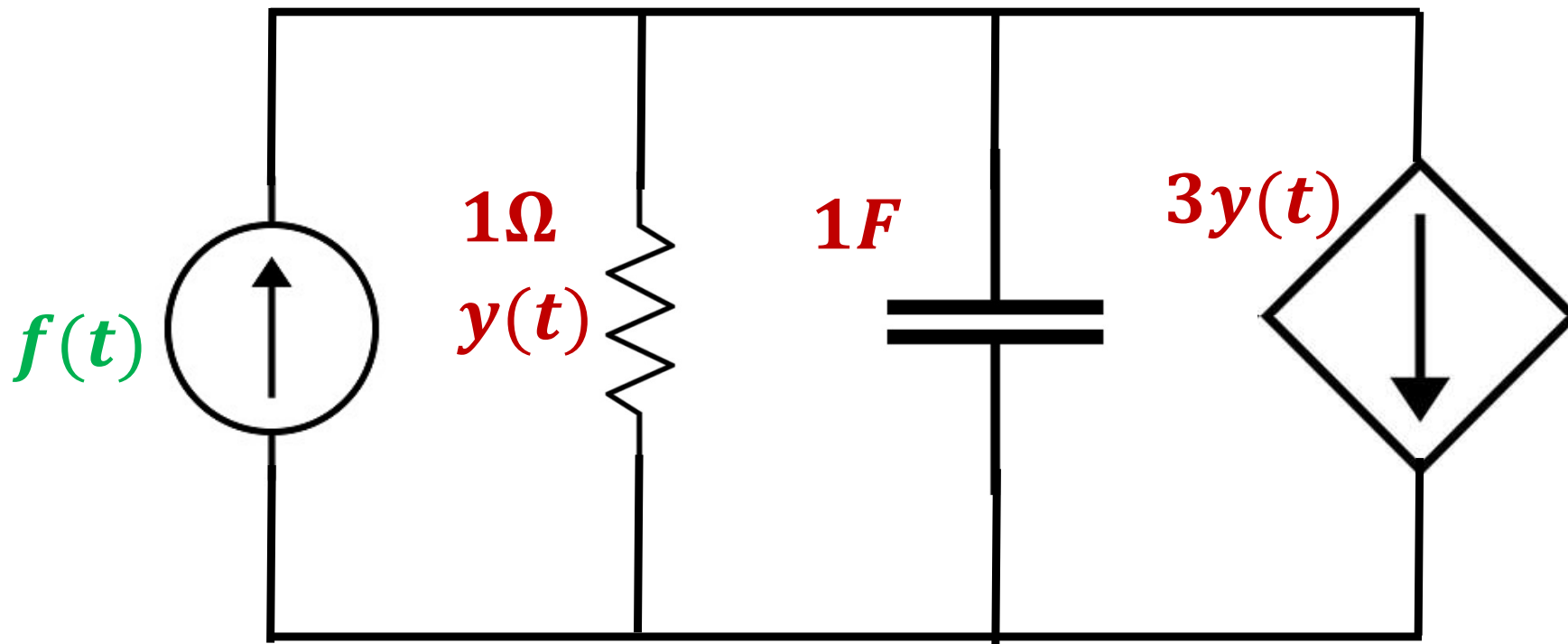
$$Z = j\omega L$$

For capacitor case, $Z = \frac{1}{j\omega C}$

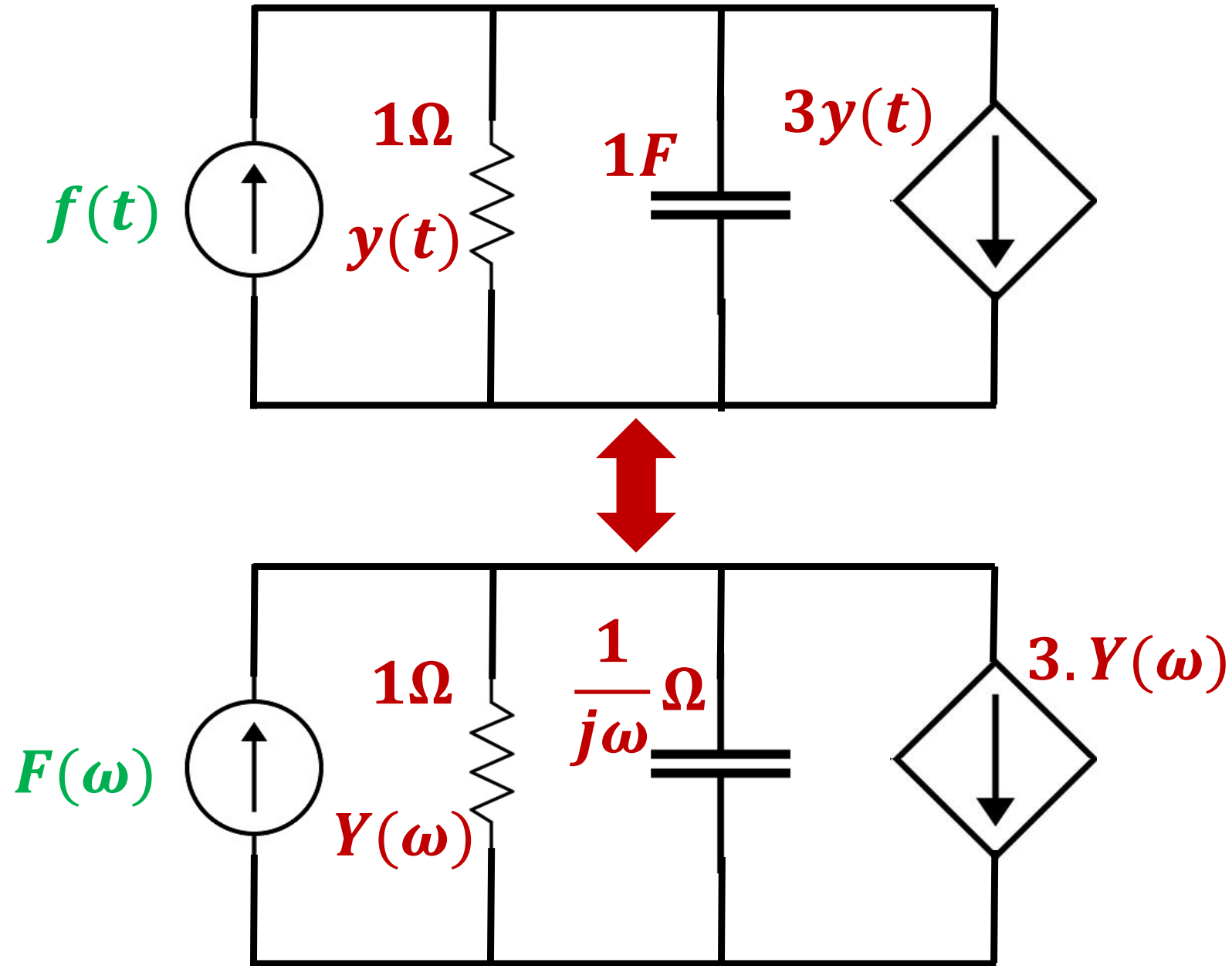
By using above, we will apply Fourier transform method to directly analyze circuits

Dissipative LTI Response – Example 7

Question: Determine the $y(t) \leftrightarrow Y(\omega)$ for an arbitrary input $f(t) \leftrightarrow F(\omega)$ for the circuit given below?



Dissipative LTI Response – Example 7



Dissipative LTI Response – Example 7

Applying node voltage method, the KCL equation for the top node is,

$$F(\omega) = \frac{Y(\omega)}{1} + \frac{Y(\omega)}{\frac{1}{j\omega}} + 3Y(\omega) = (4 + j\omega)Y(\omega)$$

Thus,

$$Y(\omega) = \frac{1}{(4+j\omega)} F(\omega)$$

which is a Fourier transform of the zero-state response $y(t)$

Dissipative LTI Response – Example 7

Hence, the system frequency response is,

$$H(\omega) = \frac{1}{(4 + j\omega)}$$

Using the inverse Fourier transform,

$$y(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} \frac{1}{4 + j\omega} F(\omega) e^{j\omega t} d\omega$$

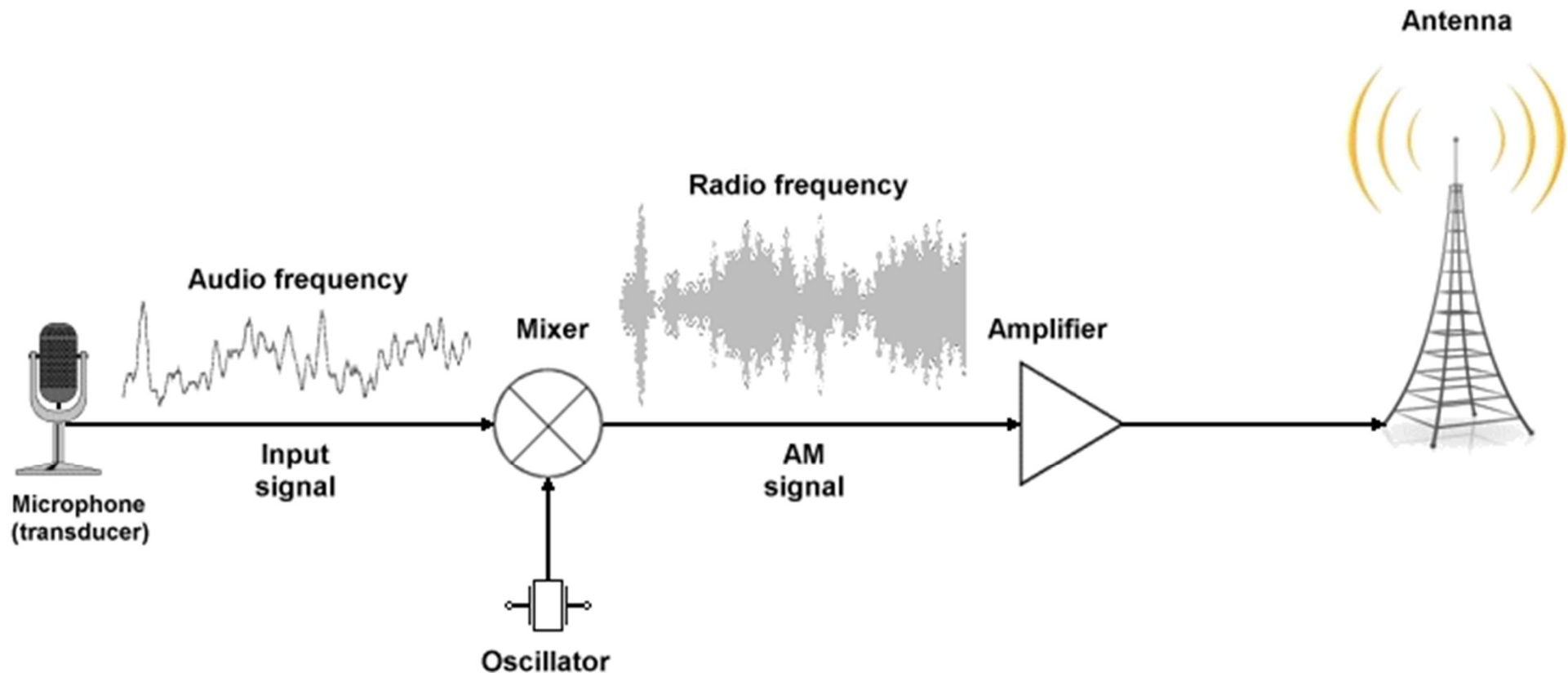
This response is valid for all lab generated input having finite energy spectrum and Fourier transformable signals

Objectives

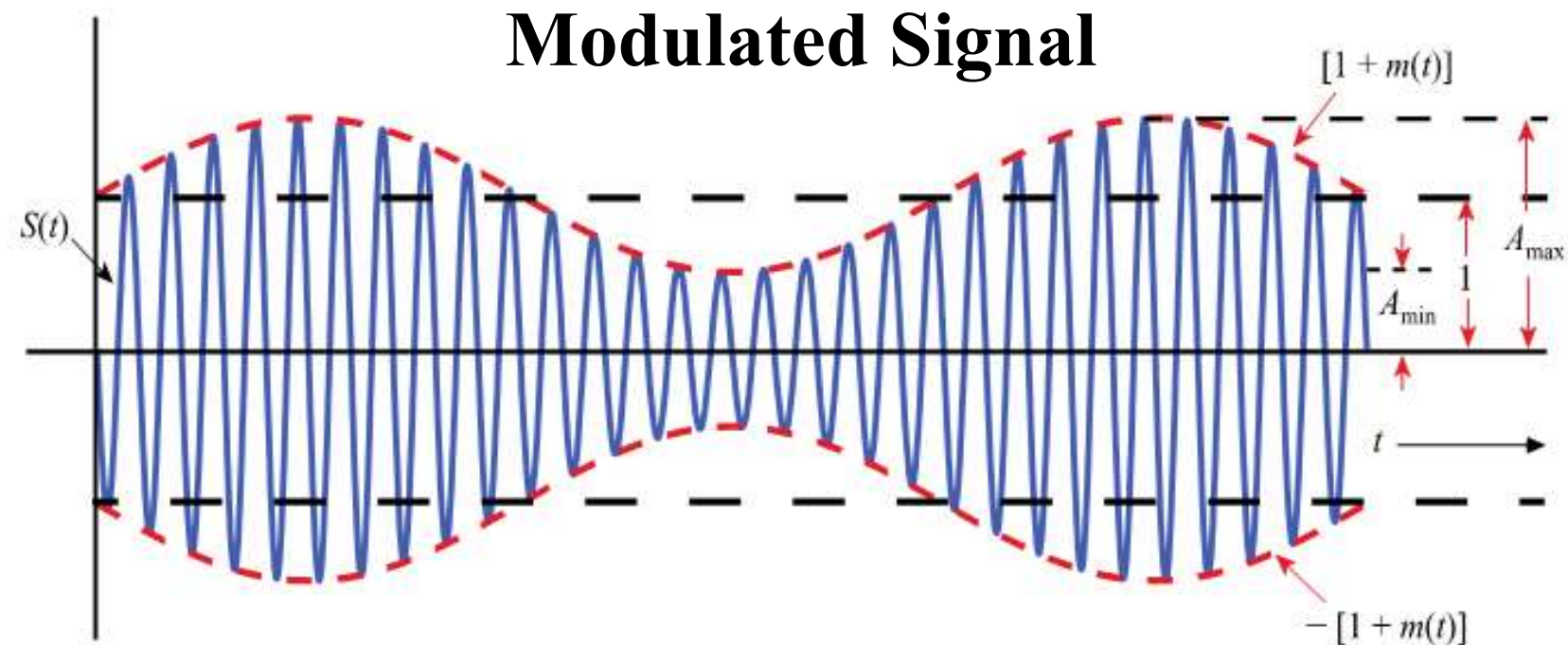
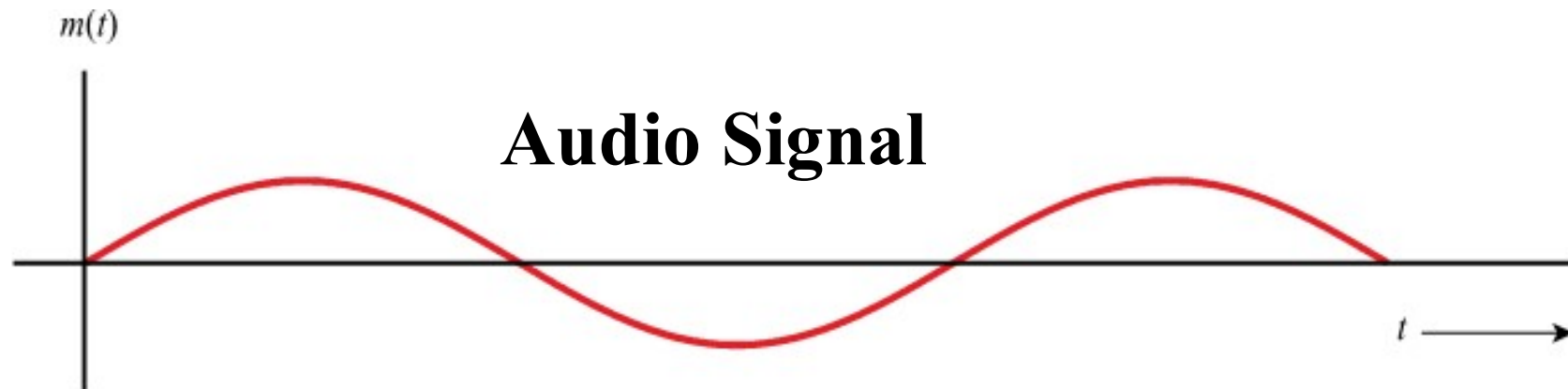
- LTI System Response to Energy Signals
- **Amplitude modulation**
- Coherent demodulation of AM signal

Amplitude Modulation

- The power limitations for voice communication at long distances evolved modulation techniques

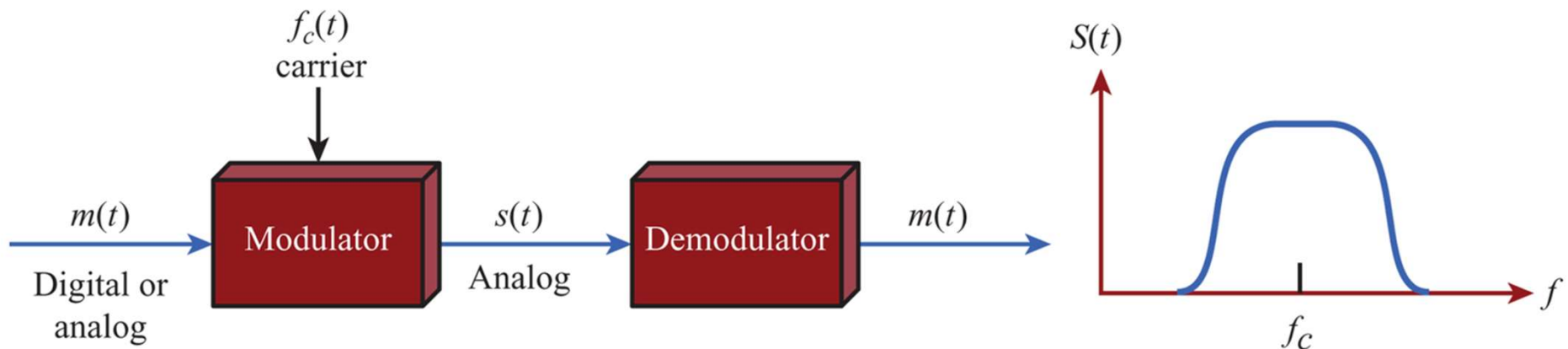


Amplitude Modulation

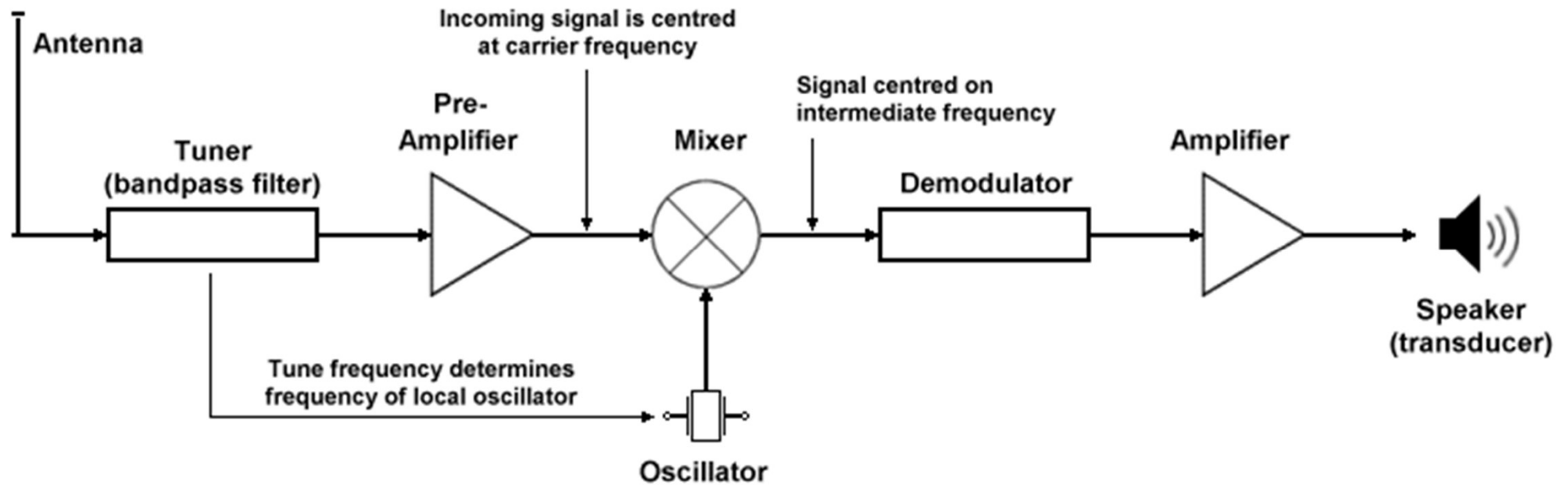


Amplitude demodulation

- The prime responsibilities of a demodulation system:
 - ✓ Receive signal power through antenna
 - ✓ Extraction of message signal from modulated signal
 - ✓ Selection of specific band spectrum (BP filter)
 - ✓ Conversion of signal into voltage/current signal
 - ✓ Passing the signal through a transducer (Headphone, speaker)



Demodulation of AM signal



Fourier properties for AM

➤ Time – Shifting Property

The time shifting property states that,

$$f(t - t_o) \longleftrightarrow F(\omega)e^{-j\omega t_o}$$

where $F(\omega)$ is the Fourier transform of $f(t)$

➤ Frequency – Shifting Property

The frequency shifting property states that,

$$F(\omega - \omega_o) \longleftrightarrow f(t)e^{jt\omega_o}$$

Fourier properties for AM – Heterodyne

- We will make use of these two properties to better understand heterodyning phenomena
- *Heterodyning* refers to translation of signal spectrum to another frequency
- The demodulating circuit needs to translate high frequency signals broadcasted by AM station to *detectable* low frequency range

Fourier properties for AM

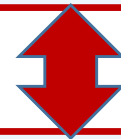
Let's consider both frequency shifting as,

$$F(\omega - \omega_c) \longleftrightarrow f(t)e^{jt\omega_c}$$

$$F(\omega + \omega_c) \longleftrightarrow f(t)e^{-jt\omega_c}$$

summing of these will lead us to,

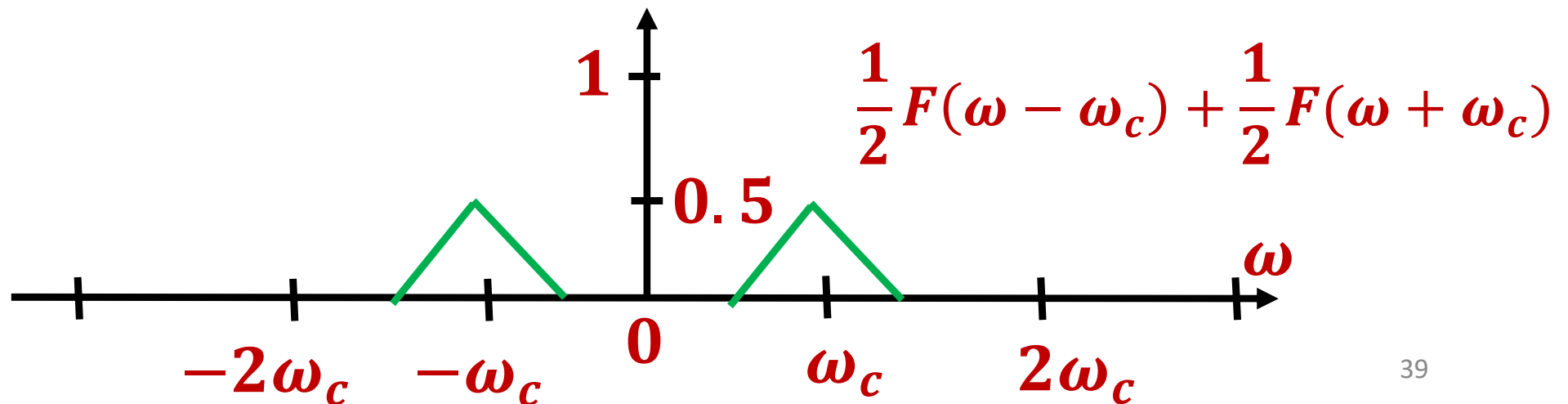
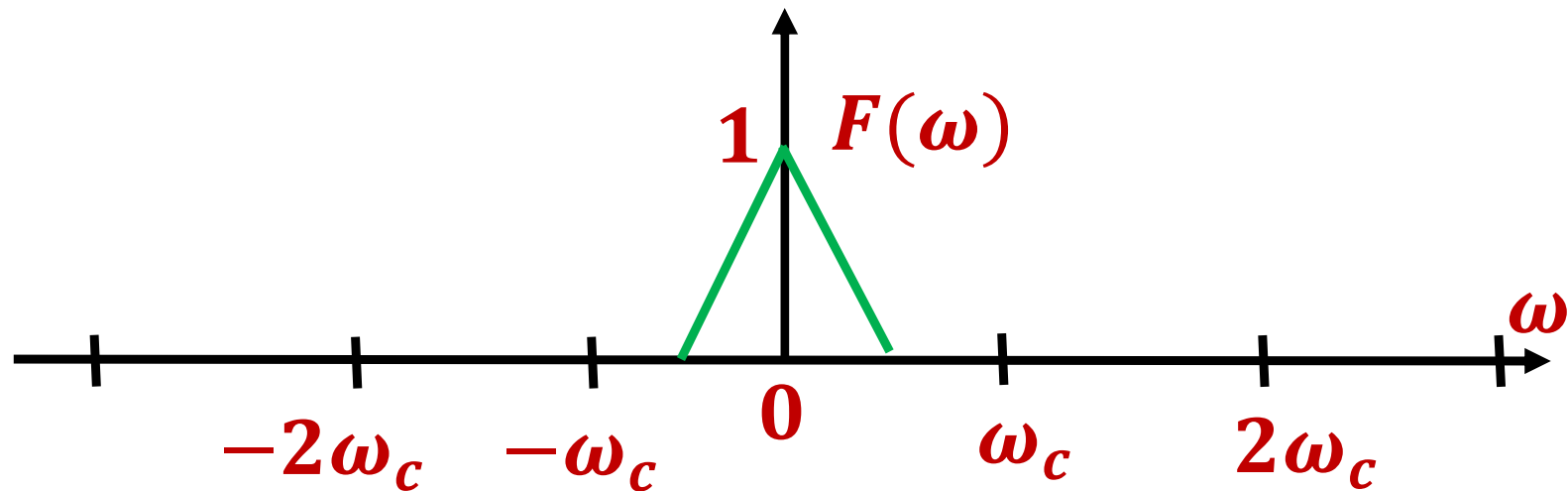
$$f(t)(e^{jt\omega_c} + e^{-jt\omega_c}) = 2 f(t)\cos(\omega_c t)$$



$$F(\omega - \omega_c) + F(\omega + \omega_c)$$

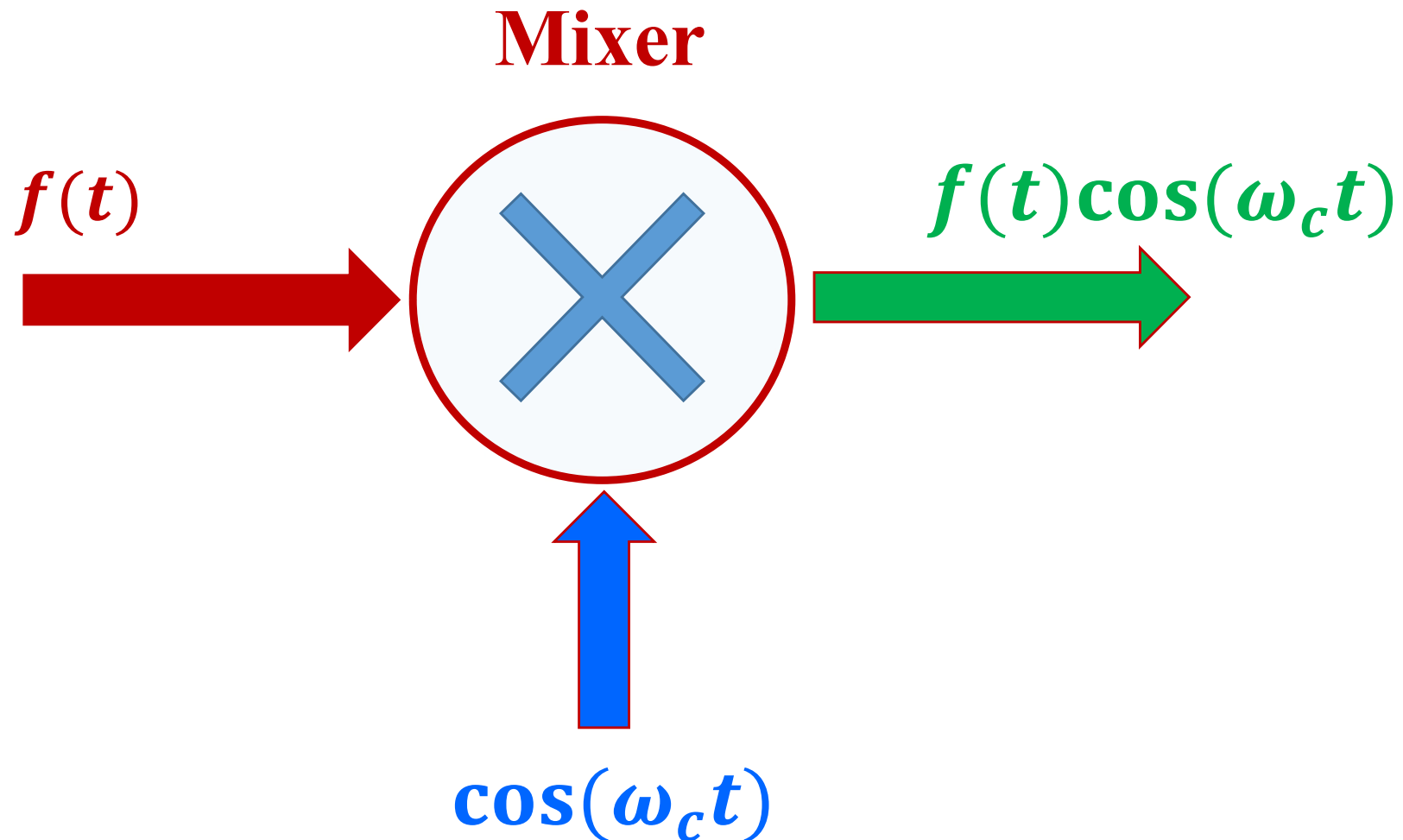
Fourier properties for AM

$$f(t)\cos(\omega_c t) \longleftrightarrow \frac{1}{2}F(\omega - \omega_c) + \frac{1}{2}F(\omega + \omega_c)$$



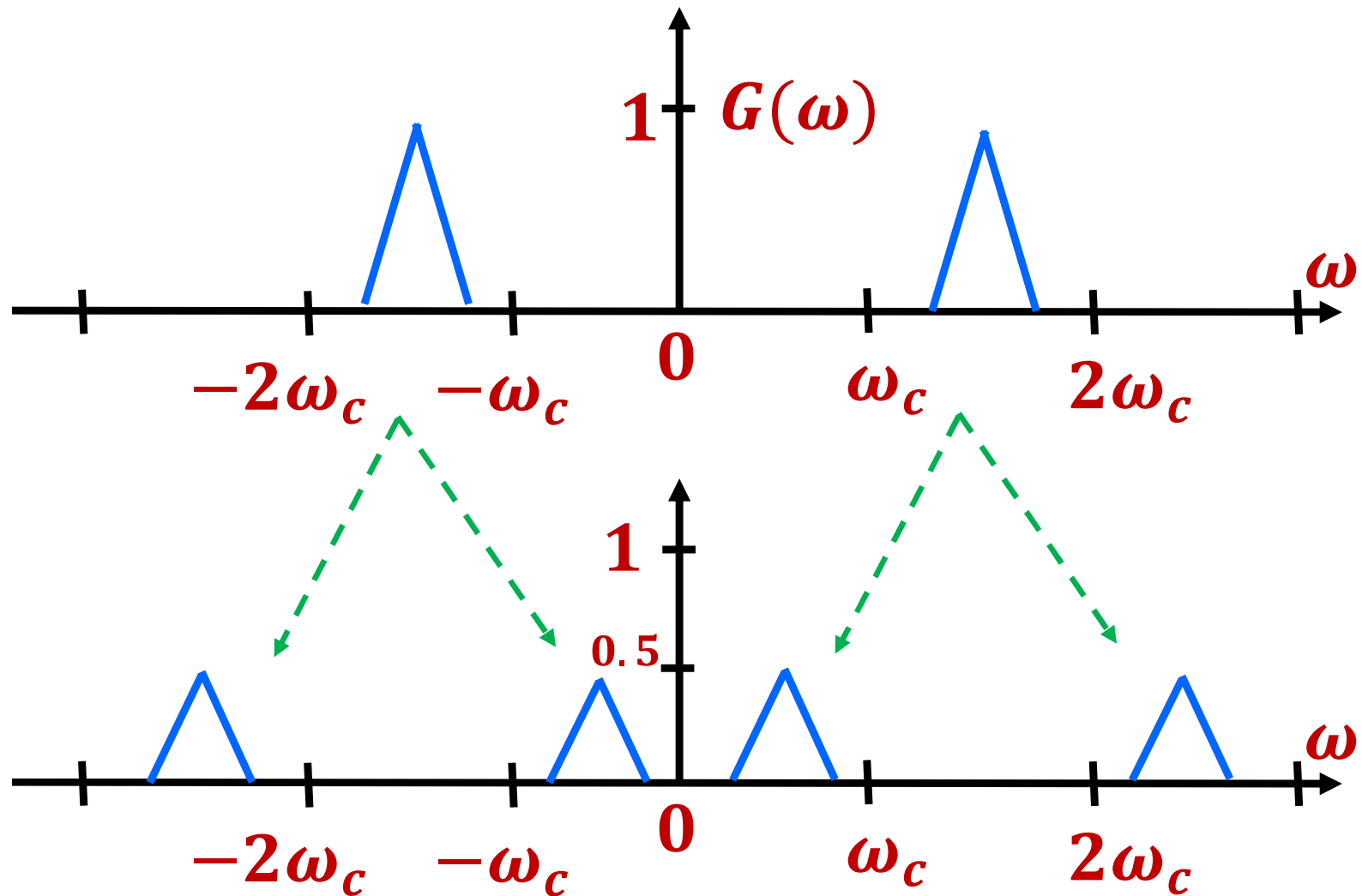
Heterodyning – Mixer

- Shifting frequency content to a new location in ω –space is called heterodyning



Another Example of modulation

$$g(t)\cos(\omega_c t) \longleftrightarrow \frac{1}{2}G(\omega - \omega_c) + \frac{1}{2}G(\omega + \omega_c)$$



Motivation for AM

- The antenna converts radio waves into electrical signal (and vice versa)
- The frequency response of antenna allows us to transmit signal at higher frequencies,

$$|H_{ant}(\omega)| = \frac{\pi c}{2L}$$

- For human voice, can you calculate antenna size L when c is the speed of light?
- Using *bandpass filtration* advantage, we can generate multiple carriers and receive the same without channel interfering problem

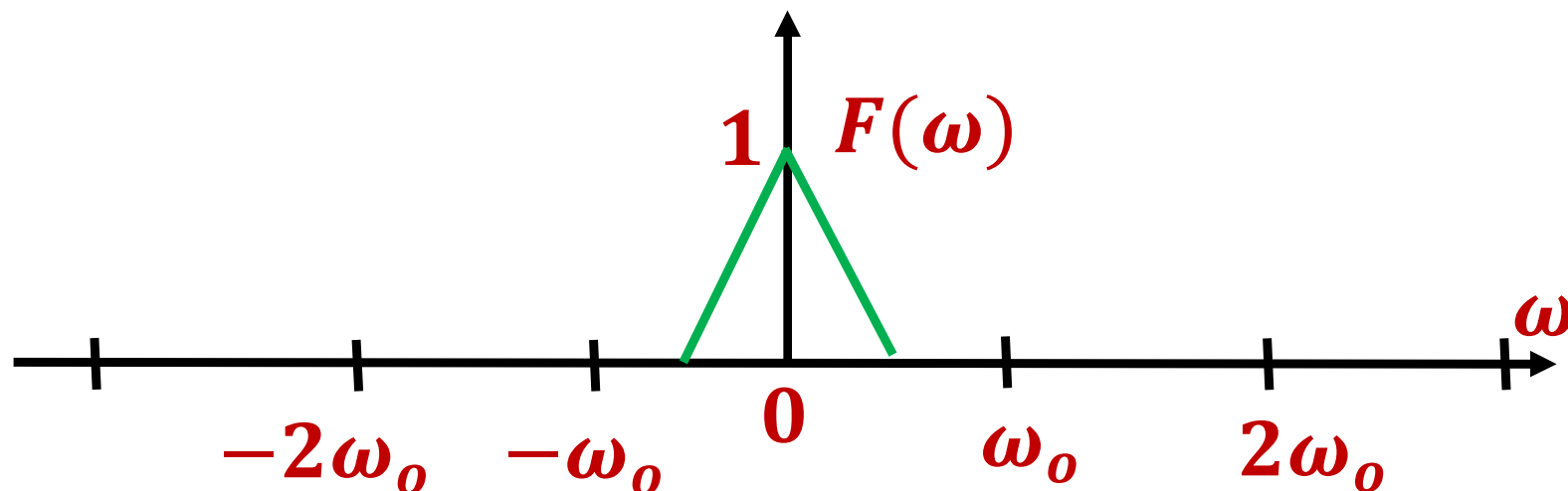
Fourier properties for AM – Example 8

Question: A mixer is used to multiply

$$1 + \sum_{n=1}^{\infty} \frac{1}{n} \cos(n\omega_o t) \quad \text{with a low pass signal } f(t)$$

plot the Fourier transform $M(\omega)$ as output of mixer?

$$m(t) = f(t) \left\{ 1 + \sum_{n=1}^{\infty} \frac{1}{n} \cos(n\omega_o t) \right\}$$



Fourier properties for AM – Example 8

Solution: We will start by expanding the expression,

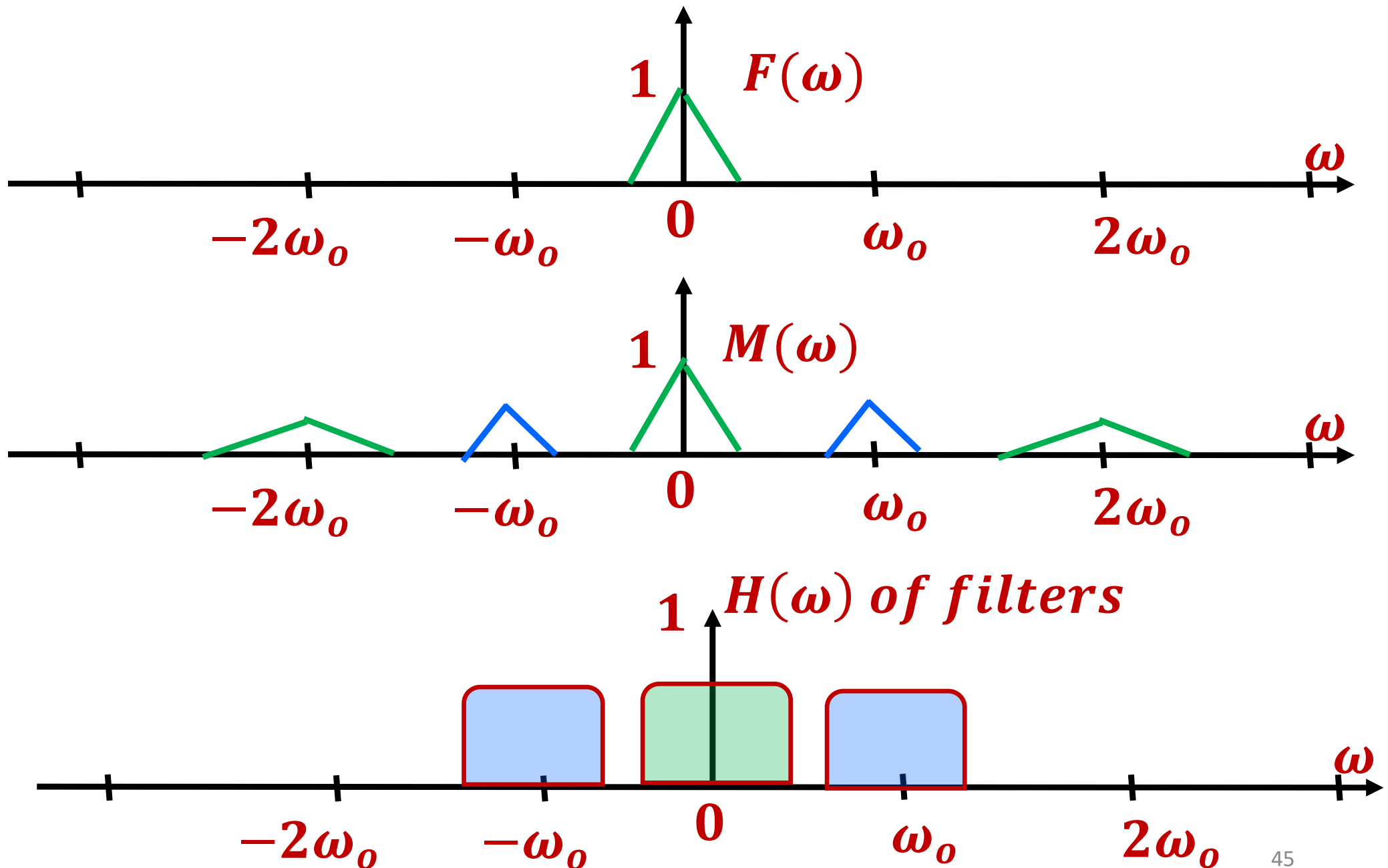
$$m(t) = f(t) + \sum_{n=1}^{\infty} \frac{1}{n} f(t) \cos(n\omega_o t)$$

taking Fourier transforms and applying Fourier transform properties gives,

$$M(\omega) = F(\omega) + \sum_{n=1}^{\infty} \frac{1}{2n} \{ F(\omega - n\omega_o) + F(\omega + n\omega_o) \}$$

Let's draw the signal spectrum and frequency responses for HPF and BPF

Fourier properties for AM – Example 8



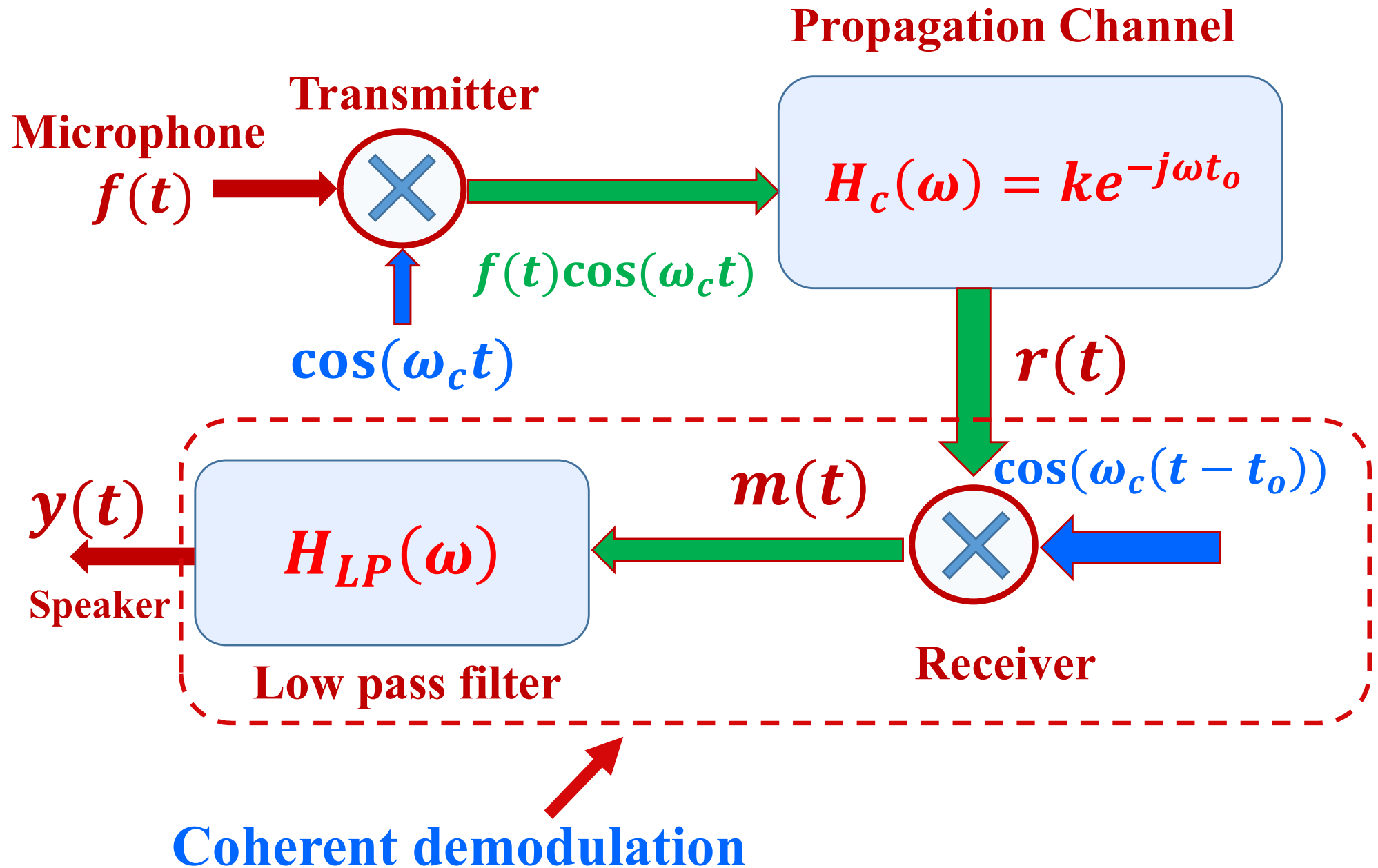
Objectives

- LTI System Response to Energy Signals
- Amplitude modulation
- **Coherent demodulation of AM signal**

Coherent demodulation

- As the transmitter and receiver are situated hundreds of kilometers away
- Any good detection should contain *rejection* component for **channel delay** and **unwanted scaling** (simply noise)
- The delay and scaling can be modelled in Fourier domain and Bandpass filter are good enough to reject unwanted frequency components in the signals
- Let's see this step by step...

Coherent demodulation –Overall picture



Coherent demodulation

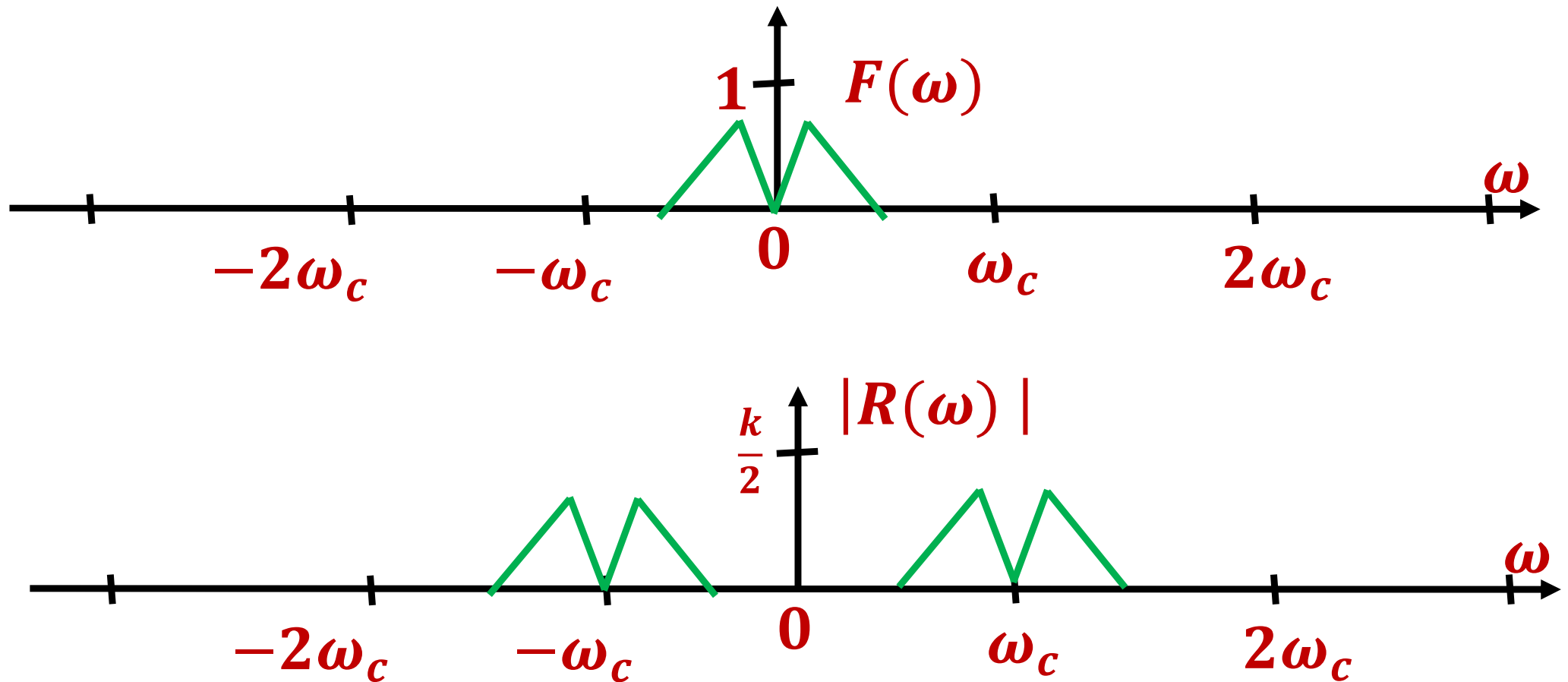
- The ideal propagation channel (delayed by t_o and scaled by k) has frequency response of,

$$H_c(\omega) = k e^{-j t_o \omega}$$

- Due to radio channel propagation losses, the signal arrived at the receiver end is delayed by t_o and scaled by k , can be written as,

$$r(t) = k f(t - t_o) \cos(\omega_c(t - t_o))$$

Coherent demodulation



Coherent demodulation

- The mixer output in the receiver is

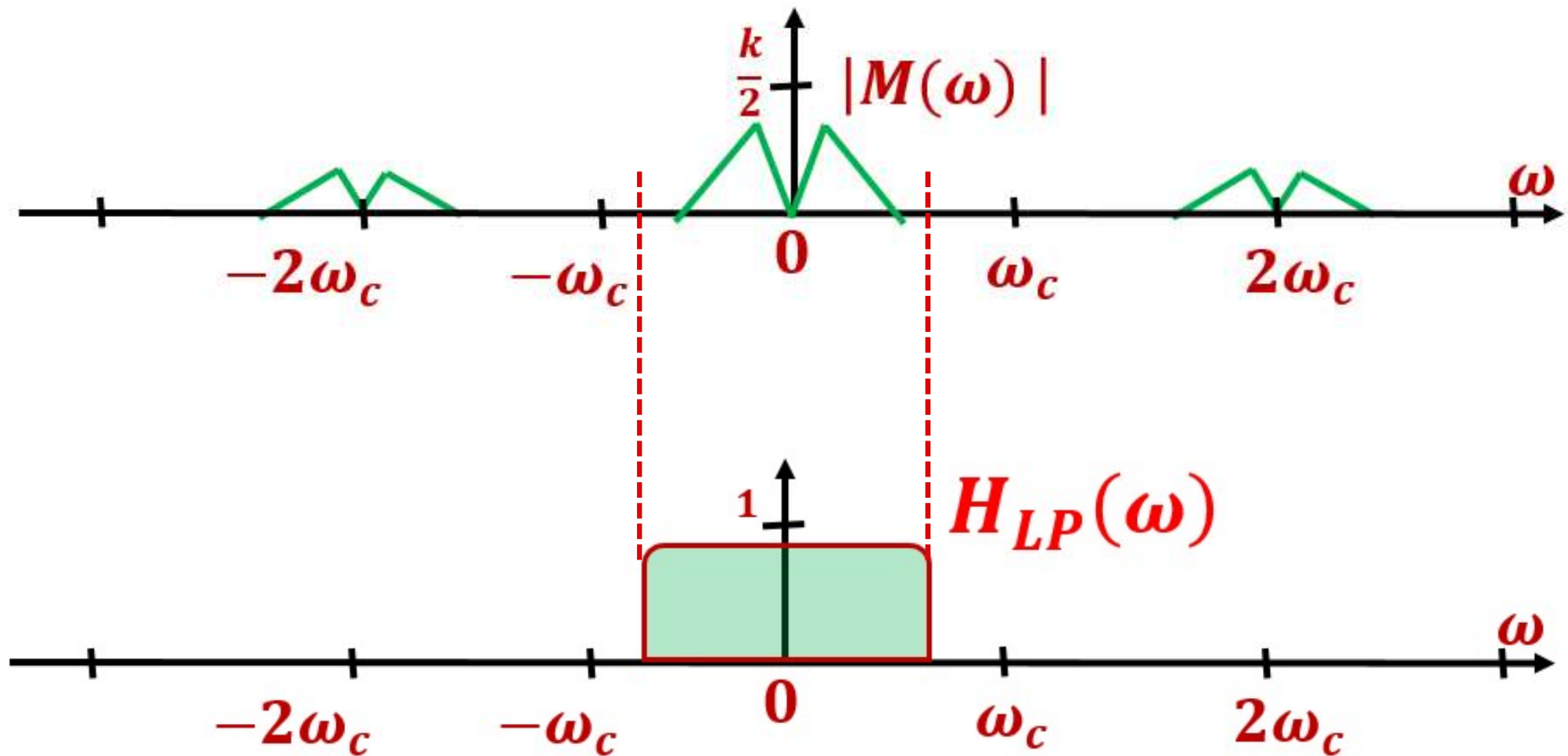
$$m(t) = r(t)\cos(\omega - \omega_o))$$

$$m(t) = f(t - t_o) \frac{k}{2} \{1 + \cos(2\omega_c(t - t_o))\}$$

- Using addition, modulation and shifting properties of Fourier transform, we can calculate $M(\omega)$ as,

$$M(\omega) = \frac{k}{2} F(\omega) e^{-j\omega t_0} + \frac{k}{4} \{F(\omega - 2\omega_c) + F(\omega + 2\omega_c)\} e^{-j\omega t_0}$$

Coherent demodulation



Coherent demodulation

➤ It can be seen that , the first term of $m(t)$,

$$\frac{k}{2} f(t - t_0)$$

is delayed audio signal we want to recover. To extract this, we pass it to a LPF ,

$$Y(\omega) = H_{LP}(\omega)M(\omega) = \frac{k}{2} F(\omega) e^{-j\omega t_0}$$

taking the inverse Fourier transform will lead us to the signal that can be input to the speaker

$$y(t) = \frac{k}{2} f(t - t_0)$$

Trouble with coherent demodulation

- The mixing of $r(t)$ and $\cos(\omega_c(t - t_o))$ at the mixer stage creates complexity as we need to generate $\cos(\omega_c t + \theta)$ locally with the right frequency ω_c
- An arbitrary phase shift $\theta \neq -\omega_c t_0$ will not work for us, as a small shift in t will generate a large *phase shift* and $y(t)$ will always fluctuate
- *Coherent*, thus, refers to the requirement that phase shift θ of incoming signal $\cos(\omega_c t + \theta)$ be coherent with phase shift $\omega_c t_0$ of incoming carrier
$$\theta \equiv -\omega_c t_0$$

How to deal with it...

- The receiver complexity can be minimized if the incoming signal is of the form,

$$r(t) = k(f(t - t_0) + \alpha)\cos(\omega_c(t - t_o))$$

- It contains a constant cosine term $k\alpha\cos(\omega_c(t - t_o))$ plus primary signal term $k(f(t - t_0))\cos(\omega_c(t - t_o))$ carrying the voice signal $f(t)$
- So, the loss can be adjusted by α and delay can be settled by t_o *locally*

Summary

- The system output is the IFT (Inverse Fourier Transform) of the product of the system frequency response and Fourier transform of the input
- In dissipative systems, transient part of zero-state response to input $\cos(\omega t)$ and $\sin(\omega t)$ applied at $t=-\infty$ must be vanished for finite times
- Once we know the system frequency response, we can estimate system output for any practical input
- The complex circuits can be directly analyzed by taking Fourier transform and evaluating frequency response

Summary

- Any energy signal can be transmitted far away by modulating it with high frequency signal
- When the modulated signal is propagated in radio channel transmitted by antenna *mixed* with channel losses and cause attenuation losses and delay in the signal
- The receiver includes demodulation circuit and filter circuits to extract message signal
- *Coherent* refers to the requirement that phase shift of incoming signal matches with the local oscillator signal

Further reading

1. Ch. 7 (page 248-258), Ch. 8 (page 261-268), E. Kudeki and D. C. Munson, *Analog Signals and Systems*, Prentice Hall, 2008.
2. Ch. 8 (page 582-593), C. K. Allan V. Oppenheim, *Signals and Systems* , 5th ed., Prentice hall, 1996.
3. Ch. 14 (page 650-670), C. K. Alexander and M. Sadiku, *Fundamentals of Electric Circuits*, 5th ed., McGraw-Hill, 2013.
4. Ch. 15 (page 755-765, page 770-785), J. D. Irwin, and R. M. Nelms. *Basic Engineering Circuit Analysis*, 10th ed. Wiley, 2011.

Preview:

1. Ch. 8 (page 268-275), E. Kudeki and D. C. Munson, *Analog Signals and Systems*, Prentice Hall, 2008.

Homework 9

Deadline: 10:00 PM, 27th April, 2022

Thank you!