



ANALOG SIGNAL PROCESSING



ECE 210 & 211
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Objectives

- **Average and rms signal power**
- **Parseval's theorem**
- **Total harmonic distortion**
- **Fourier Transforms**
- **Properties of Fourier transform**
- **Circuit interpretation for Fourier transforms**
- **Signal Energy and Spectrum**
- **Signal Bandwidth**

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Average Signal Power

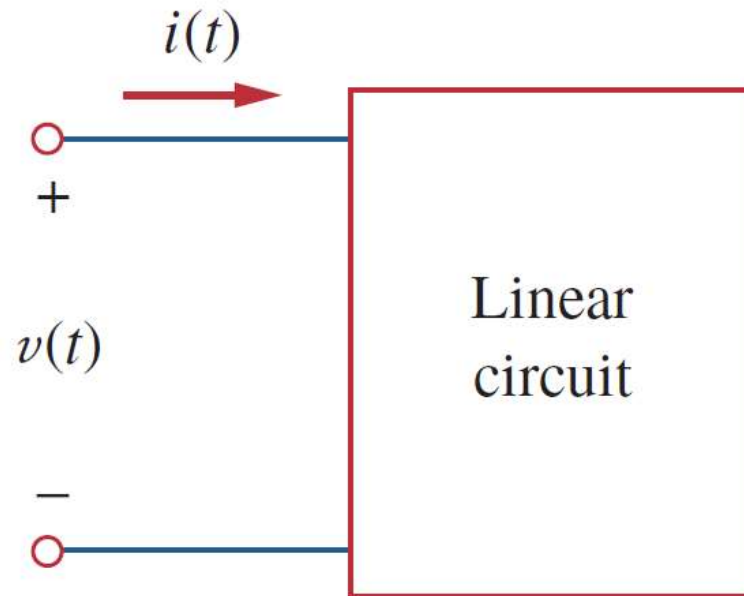
- To find the average power absorbed by a circuit due to **a periodic excitation**, we write the voltage and current in amplitude-phase form

$$v(t) = V_{\text{dc}} + \sum_{n=1}^{\infty} V_n \cos(n\omega_0 t - \theta_n)$$

$$i(t) = I_{\text{dc}} + \sum_{m=1}^{\infty} I_m \cos(m\omega_0 t - \phi_m)$$

Average Signal Power

- Recall that for a passive sign convention, we will use following direction vs sign notation as given below



- The average power is,

$$P = \frac{1}{T} \int_0^T v i \, dt$$

Average Signal Power

Substituting the values of v and i gives 4 terms as,

$$\begin{aligned} P = & \underbrace{\frac{1}{T} \int_0^T V_{\text{dc}} I_{\text{dc}} dt}_{\text{1}} + \underbrace{\sum_{m=1}^{\infty} \frac{I_m V_{\text{dc}}}{T} \int_0^T \cos(m\omega_0 t - \phi_m) dt}_{\text{2}} \\ & + \underbrace{\sum_{n=1}^{\infty} \frac{V_n I_{\text{dc}}}{T} \int_0^T \cos(n\omega_0 t - \theta_n) dt}_{\text{3}} \\ & + \underbrace{\sum_{m=1}^{\infty} \sum_{n=1}^{\infty} \frac{V_n I_m}{T} \int_0^T \cos(n\omega_0 t - \theta_n) \cos(m\omega_0 t - \phi_m) dt}_{\text{4}} \end{aligned}$$

Average Signal Power

- The 2nd and 3rd term will be *vanished* due to integration over period of cosine
- The 4th term will also be zero when **m ≠ n**
- Evaluating 1st term gives,

$$P = V_{dc}I_{dc} + \frac{1}{2} \sum_{n=1}^{\infty} V_n I_n \cos(\theta_n - \phi_n)$$

- The total average power is the sum of the average powers in each harmonically related voltage and current

Average Signal Power

The rms value of a periodic function $f(t)$ is given by,

$$F_{\text{rms}} = \sqrt{\frac{1}{T} \int_0^T f^2(t) dt} \quad \text{Eq. 1}$$

Consider $f(t)$ as,

$$f(t) = a_0 + \sum_{n=1}^{\infty} A_n \cos(n\omega_0 t + \phi_n)$$

substituting the value of $f(t)$ into **Eq. 1** gives,

Average Signal Power

$$\begin{aligned} F_{\text{rms}}^2 &= \frac{1}{T} \int_0^T \left[a_0^2 + 2 \sum_{n=1}^{\infty} a_0 A_n \cos(n\omega_0 t + \phi_n) \right. \\ &\quad \left. + \sum_{n=1}^{\infty} \sum_{m=1}^{\infty} A_n A_m \cos(n\omega_0 t + \phi_n) \cos(m\omega_0 t + \phi_m) \right] dt \\ &= \frac{1}{T} \int_0^T a_0^2 dt + 2 \sum_{n=1}^{\infty} a_0 A_n \frac{1}{T} \int_0^T \cos(n\omega_0 t + \phi_n) dt \\ &\quad + \sum_{n=1}^{\infty} \sum_{m=1}^{\infty} A_n A_m \frac{1}{T} \int_0^T \cos(n\omega_0 t + \phi_n) \cos(m\omega_0 t + \phi_m) dt \end{aligned}$$

Average Signal Power

Using the same reasoning used earlier for 2nd -4th terms and simplification gives,

$$F_{\text{rms}}^2 = a_0^2 + \frac{1}{2} \sum_{n=1}^{\infty} A_n^2 \longrightarrow F_{\text{rms}} = \sqrt{a_0^2 + \frac{1}{2} \sum_{n=1}^{\infty} A_n^2}$$

in terms of Fourier coefficients, it can be written as,

$$F_{\text{rms}} = \sqrt{a_0^2 + \frac{1}{2} \sum_{n=1}^{\infty} (a_n^2 + b_n^2)}$$

Average Signal Power

If $f(t)$ is the current through a resistor R , then the power dissipated in the resistor is

$$P = RF_{\text{rms}}^2$$

Or if $f(t)$ is the voltage across a resistor R , the power dissipated in the resistor is

$$P = \frac{F_{\text{rms}}^2}{R}$$

The power dissipated by the $1\text{-}\Omega$ resistance

$$P_{1\Omega} = F_{\text{rms}}^2 = a_0^2 + \frac{1}{2} \sum_{n=1}^{\infty} (a_n^2 + b_n^2)$$

Objectives

- Average and rms signal power
- **Parseval's theorem**
- Total harmonic distortion
- Fourier Transforms
- Properties of Fourier transform
- Circuit interpretation for Fourier transforms
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- Signal Bandwidth

Average Power – Parseval's Theorem

- The average power in a periodic signal is the sum of the average power in its dc component and the average powers in its harmonics

$$P = F_{\text{rms}}^2 = a_0^2 + \frac{1}{2} \sum_{n=1}^{\infty} (a_n^2 + b_n^2)$$

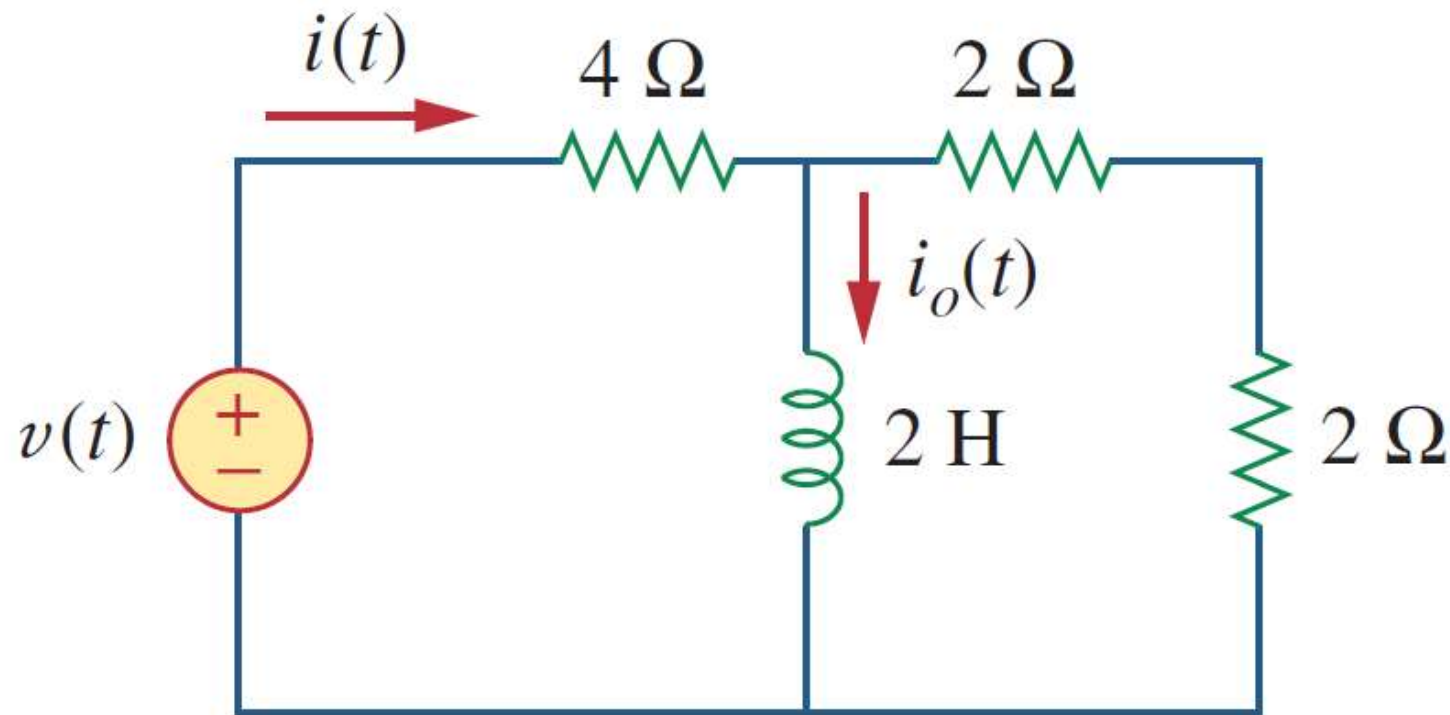
Power in the dc component

AC power in the n -th harmonic

Average Signal Power – Example 1

Question: Find the response $i_o(t)$ if the input $v(t)$ is given by *Fourier series expansion* as,

$$v(t) = 1 + \sum_{n=1}^{\infty} \frac{2(-1)^n}{1+n^2} (\cos nt - n \sin nt)$$



Average Signal Power – Example 1

Solution: We can express the input voltage as,

$$\begin{aligned} v(t) &= 1 + \sum_{n=1}^{\infty} \frac{2(-1)^n}{\sqrt{1+n^2}} \cos(nt + \tan^{-1} n) \\ &= 1 - 1.414 \cos(t + 45^\circ) + 0.8944 \cos(2t + 63.45^\circ) \\ &\quad - 0.6345 \cos(3t + 71.56^\circ) - 0.4851 \cos(4t + 78.7^\circ) + \dots \end{aligned}$$

We notice that $\omega_o = 1$, and $\omega_n = n \text{ rad/s}$, the impedance at the source is,

$$Z = 4 + j\omega_n 2 \parallel 4 = \frac{8 + j\omega_n 8}{2 + j\omega_n}$$

Average Signal Power – Example 1

The input current is,

$$I = \frac{V}{Z} = \frac{2 + j\omega_n}{8 + j\omega_n 8} V$$

where V is the phasor form of the source voltage $v(t)$

By using current division,

$$I_o = \frac{4}{4 + j\omega_n 2} I = \frac{V}{4 + j\omega_n 4}$$

since, $\omega_n = n$, I_o can be expressed as,

$$I_o = \frac{V}{4\sqrt{1 + n^2} \angle \tan^{-1} n}$$

for dc component, $n = 0$ so, $\omega_n = 0$, So, $V = 1$, $I_o = \frac{1}{4}$

Average Signal Power – Example 1

for n -th harmonic,

$$V = \frac{2(-1)^n}{\sqrt{1+n^2}} \angle \tan^{-1} n$$

so that,

$$\begin{aligned} I_o &= \frac{1}{4\sqrt{1+n^2} \angle \tan^{-1} n} \frac{2(-1)^n}{\sqrt{1+n^2}} \angle \tan^{-1} n \\ &= \frac{(-1)^n}{2(1+n^2)} \end{aligned}$$

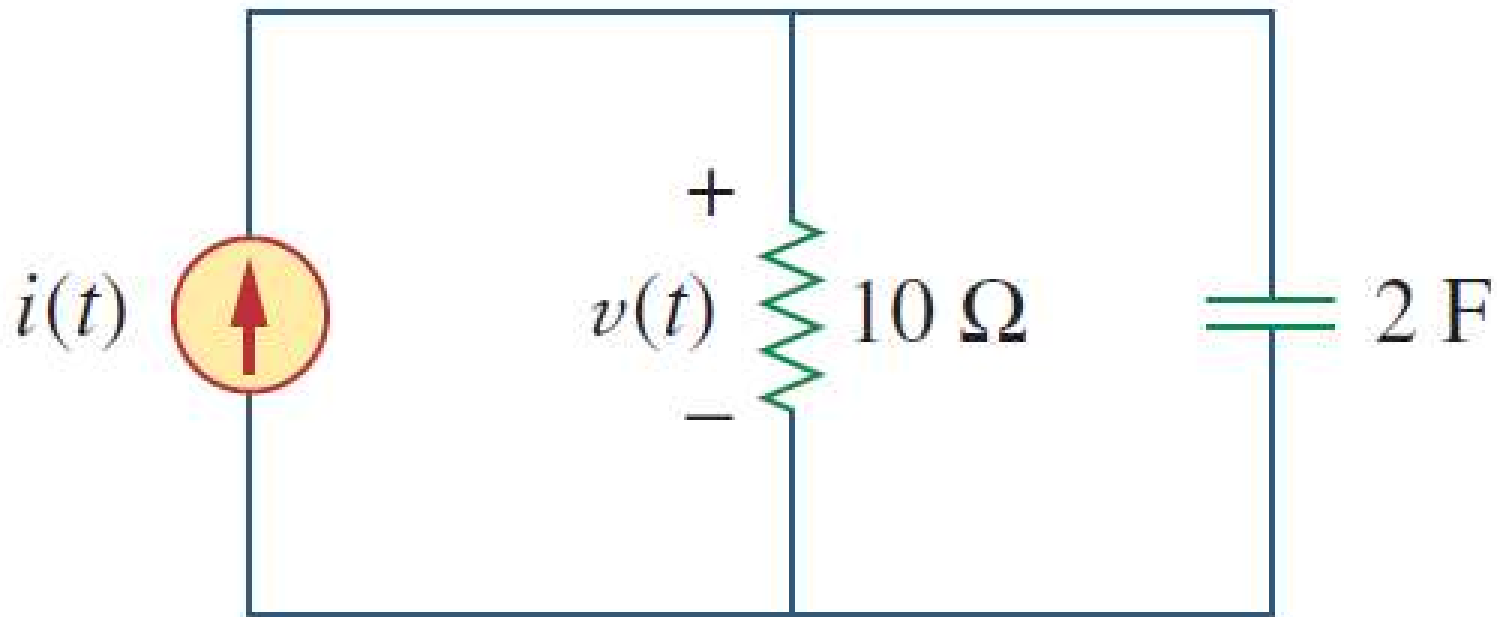
in the time domain,

$$i_o(t) = \frac{1}{4} + \sum_{n=1}^{\infty} \frac{(-1)^n}{2(1+n^2)} \cos nt A$$

Average Signal Power – Example 2

Question: Determine the average power supplied to the circuit if,

$$i(t) = 2 + 10 \cos(t + 10^\circ) + 6 \cos(3t + 35^\circ) \text{ A}$$



Average Signal Power – Example 2

Solution: The input impedance of the network is,

$$Z = 10 \parallel \frac{1}{j2\omega} = \frac{10 \left(\frac{1}{j2\omega} \right)}{10 + \frac{1}{j2\omega}} = \frac{10}{1 + j20\omega}$$

Hence,

$$V = IZ = \frac{10I}{\sqrt{1 + 400\omega^2} \angle \tan^{-1} 20\omega}$$

Average Signal Power – Example 2

For dc component, $\omega = 0$,

$$I = 2 \text{ A} \quad \text{and} \quad V = 10(2) = 20 \text{ V}$$

Expected, as the capacitor is an *open circuit* to DC and the entire 2-A current flows through the resistor

For $\omega = 1 \text{ rad/s}$,

$$I = 10 \angle 10^\circ \text{ A}, \quad V = \frac{10(10 \angle 10^\circ)}{\sqrt{1 + 400\omega^2} \angle \tan^{-1} 20}$$
$$= 5 \angle -77.1^\circ$$

Average Signal Power – Example 2

For $\omega = 3 \text{ rad/s}$,

$$\begin{aligned} I &= 6\angle 35^\circ \text{ A}, & V &= \frac{10(6\angle 35^\circ)}{\sqrt{1 + 3600\omega^2 \angle \tan^{-1} 60}} \\ & & &= 1\angle -54^\circ \end{aligned}$$

Converting into time domain,

$$v(t) = 20 + 5 \cos(t - 77.1^\circ) + 1 \cos(3t - 54^\circ) \text{ V}$$

Average Signal Power – Example 2

We obtain the average power supplied to the circuit by applying Eq.

$$P = V_{\text{dc}}I_{\text{dc}} + \frac{1}{2} \sum_{n=1}^{\infty} V_n I_n \cos(\theta_n - \phi_n)$$

To get the proper signs of θ_n and ϕ_n , we will compare with values of v and i ,

$$\begin{aligned} P &= 2(20) + \frac{1}{2}(5)(10)\cos[77.1^\circ - (-10^\circ)] \\ &\quad + \frac{1}{2}(1)(6)\cos[54.0^\circ - (-35^\circ)] \\ &= 40 + 1.24 + 0.005 = 41.6 \text{ W} \end{aligned}$$

Average Signal Power – Example 3

Question: From Example 1, estimate the rms value of voltage given as,

$$\begin{aligned} v(t) = & 1 - 1.414 \cos(t + 45^\circ) + 0.89 \cos(2t + 63.45^\circ) \\ & - 0.634 \cos(3t + 71.55^\circ) \\ & - 0.48 \cos(4t + 78.7^\circ) + \dots V \end{aligned}$$

Average Signal Power – Example 3

Solution:

Using the following relation to estimate rms value of the voltage $v(t)$

$$V_{rms} = \sqrt{a_o^2 + \frac{1}{2} \sum_{n=1}^{\infty} A_n^2}$$

$$V_{rms} = \sqrt{1^2 + \frac{1}{2} [(-1.41)^2 + (0.89)^2 + (-0.63)^2 + (-0.48)^2 + \dots]}$$

$$V_{rms} = 1.649 \text{ V}$$

Average Signal Power – Example 3

- This is only an estimate, as we have not taken enough terms of the series
- The actual function represented by the Fourier series is

$$v(t) = \frac{\pi e^t}{\sinh \pi}, \quad -\pi < t < \pi$$

with $v(t) = v(t + T)$,

The exact rms value of this is 1.776 V

Objectives

- Average and rms signal power
- Parseval's theorem
- **Total harmonic distortion**
- Fourier Transforms
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Simple Harmonic Distortions

Suppose a system with $y(t)$ as response is written as,

$$y(t) = Af(t) + Bf^2(t)$$

Linear Response

Desired



Nonlinear Response

Undesired



How to estimate the effects of undesired part of response!

Simple Harmonic Distortions

We know that

- A linear system will respond a **cosine input** with a **cosine output** at *same frequency*
- But if **output generates** *higher order frequency*

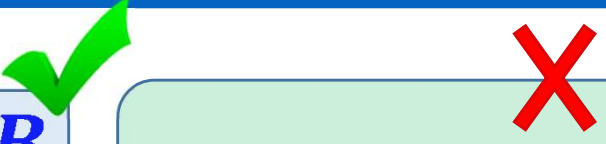
$$f(t) = \cos(\omega_o t)$$

as,

$$y(t) = A\cos(\omega_o t) + B\cos^2(\omega_o t)$$

$$y(t) = \frac{B}{2} + A\cos(\omega_o t) + \frac{B}{2}\cos(2\omega_o t)$$

Simple Harmonic Distortion

$$y(t) = \boxed{\frac{B}{2}} + \boxed{A \cos(\omega_o t) + \frac{B}{2} \cos(2\omega_o t)}$$


- Clearly, you can see undesired harmonic part (DC term *plus* harmonic) is added into purely cosine input
- According to *Parseval's theorem*, average power is,

$$P = F_{\text{rms}}^2 = \boxed{a_0^2} + \boxed{\frac{1}{2} \sum_{n=1}^{\infty} (a_n^2 + b_n^2)}$$

DC Component

AC power in the n -th harmonic

Simple Harmonic Distortion

$$P_y = \frac{B^2}{4} + \frac{A^2}{2} + \frac{B^2}{8}$$

DC Component

Average power of fundamental and second harmonic

- One important consideration is *simple or second harmonic distortion*
- It can be achieved by taking ratio of average power in second harmonic and to the average power in fundamental,

$$S.H.D = \frac{B^2}{8} : \frac{A^2}{2} = \frac{B^2}{4A^2}$$

Total Harmonic Distortion

- More generally, a non-linear system may respond to a pure cosine input using Fourier series as,

$$y(t) = \frac{c_0}{2} + \sum_{n=1}^{\infty} c_n \cos(n\omega_o t + \theta_n)$$

- Contains fundamental *plus* infinite higher order harmonics
- It's sensible to consider most affective harmonics in the calculation of overall distortion

Total Harmonic Distortion

- So, we take ratio of **second and higher order harmonics** to the **fundamental harmonic** is called *total harmonic distortion*

$$T.H.D = \frac{\sum_{n=2}^{\infty} \frac{1}{2} c_n^2}{\frac{c_1^2}{2}} = \frac{\sum_{n=2}^{\infty} c_n^2}{c_1^2}$$

Second and higher order harmonics = $\sum_{n=2}^{\infty} \frac{1}{2} c_n^2$

From the signal,

$$y(t) = \frac{c_0}{2} + \sum_{n=1}^{\infty} c_n \cos(n\omega_o t + \theta_n)$$

Total Harmonic Distortion

Second and higher order harmonics = *Average Signal power* – (average power of first and fundamental harmonics)

$$\sum_{n=2}^{\infty} c_n^2 = 2 \left(P_y - \left(\frac{c_o^2}{4} + \frac{1}{2} c_1^2 \right) \right)$$

Putting the value,

$$T.H.D = \frac{P_y - \left(\frac{c_o^2}{4} + \frac{1}{2} c_1^2 \right)}{\frac{1}{2} c_1^2}$$

Where P_y is the average signal power of $y(t)$

Total Harmonic Distortion – Example 4

Question: A system is supposed to deliver ,

$$y(t) = \cos(\omega_o t) \text{ @ } \frac{\omega_o}{2\pi} = 60\text{Hz}$$

But, it actually delivers,

$$y(t) = \cos(\omega_o t) + \frac{1}{9} \cos(3\omega_o t) + \frac{1}{25} \cos(5\omega_o t)$$

What is the Total Harmonic Distortion (T.H.D) in the system?

Total Harmonic Distortion – Example 4

Solution:

- It can be seen that taking *ratio* of second and third harmonic to the fundamental harmonic gives the *T.H.D*,
- As per Parseval's theorem,

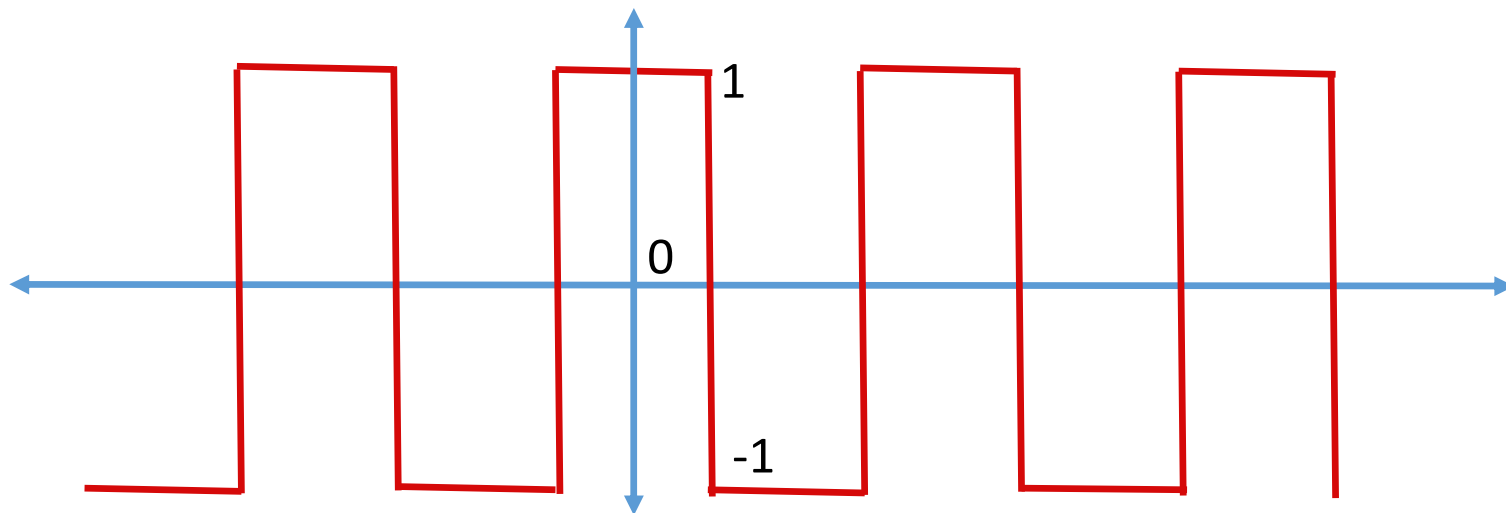
$$T.H.D = \frac{\left(\frac{1}{9}\right)^2 + \left(\frac{1}{25}\right)^2}{1^2} \approx 1.40\%$$

Total Harmonic Distortion – Example 5

Question: The Fourier of zero-mean-square wave is given by,

$$y(t) = \frac{4}{\pi} \left[\cos(t) - \frac{1}{3} \cos(3t) + \frac{1}{5} \cos(5t) - \frac{1}{7} \cos(7t) + \dots \right]$$

Assume that $y(t)$ is proportional to $\cos(t)$, find total harmonic distortion of the signal?



Total Harmonic Distortion – Example 5

Solution: Since the signal has zero mean, so,

$$c_0^2 = 0, \quad \text{also,} \quad \frac{c_1^2}{2} = \frac{8}{\pi^2}$$

Taking the average power of signal over entire period gives,

$$P_y = \frac{1}{2\pi} \int_0^{2\pi} 1^2 dt = 1$$

Using T.H.D relation,

$$T.H.D = \frac{1 - \frac{8}{\pi^2}}{\frac{8}{\pi^2}} \approx 23.4\%$$

Objectives

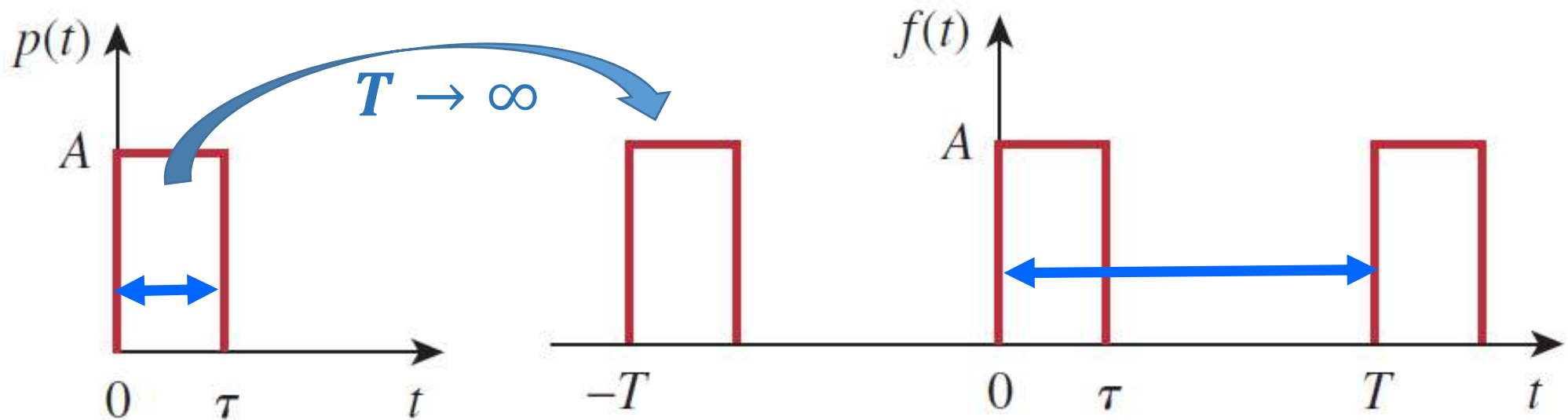
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Fourier Transforms

- **A non-sinusoidal periodic function can be represented by a Fourier series, provided that it satisfies the Dirichlet conditions**
- **Non-Periodic function is defined as a signal having non-repeating pattern in range $[-\infty, \infty]$**
- **What if the signal is non-periodic?**
- **Is Fourier series exist for non-periodic signals**

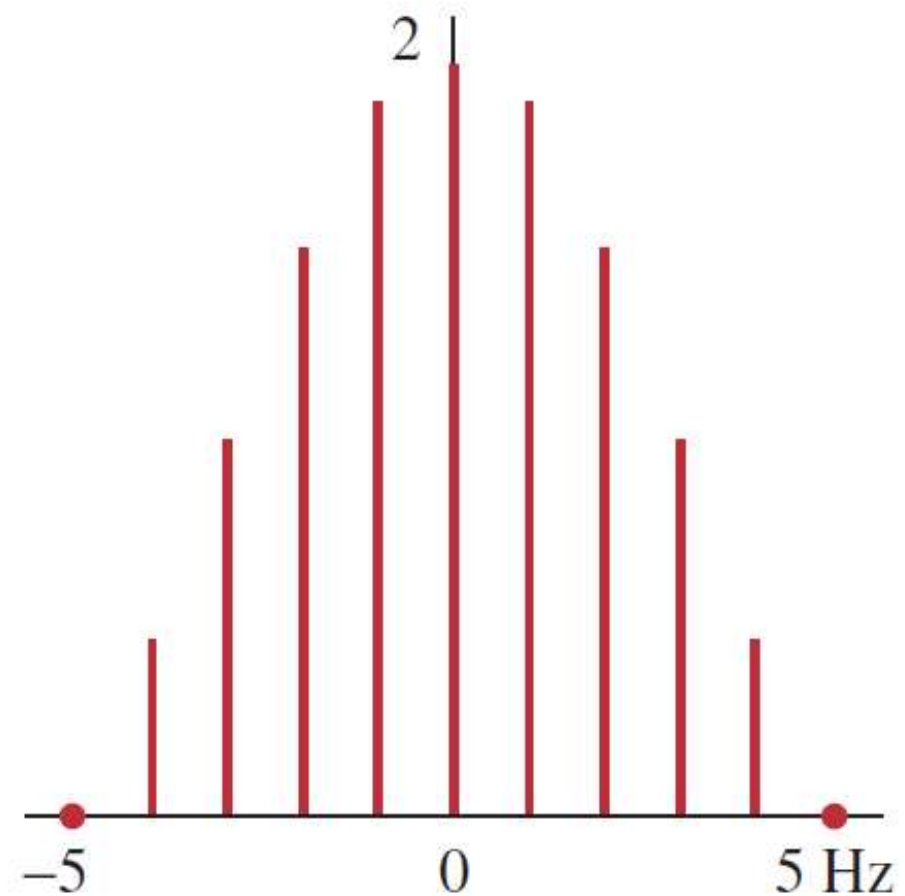
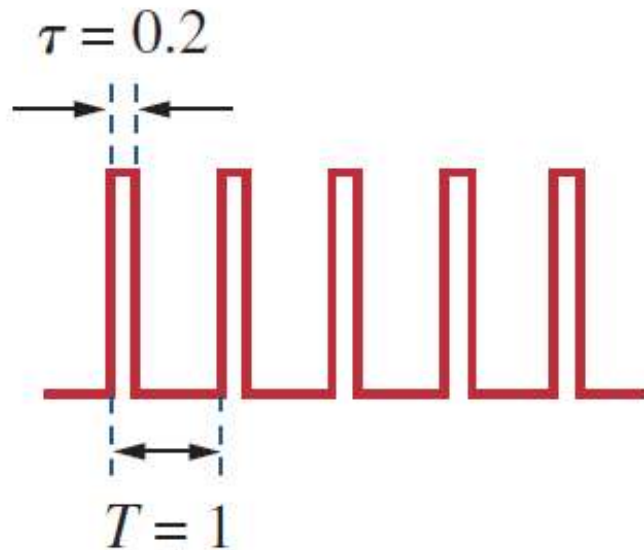
Fourier Transforms

- The answer is – Yes!
- Suppose a non-periodic function whose period extends up to ∞



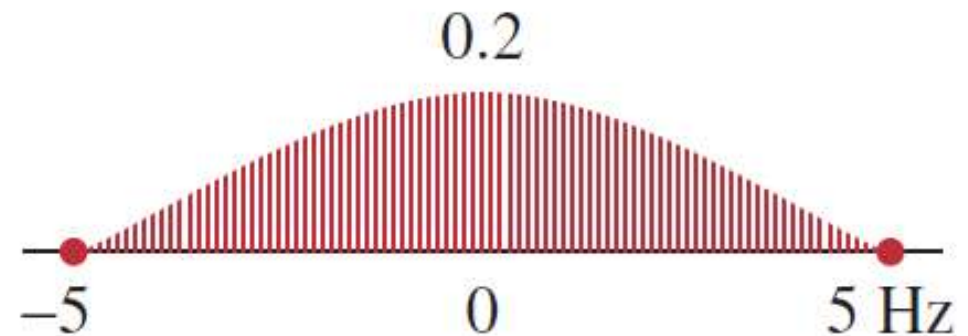
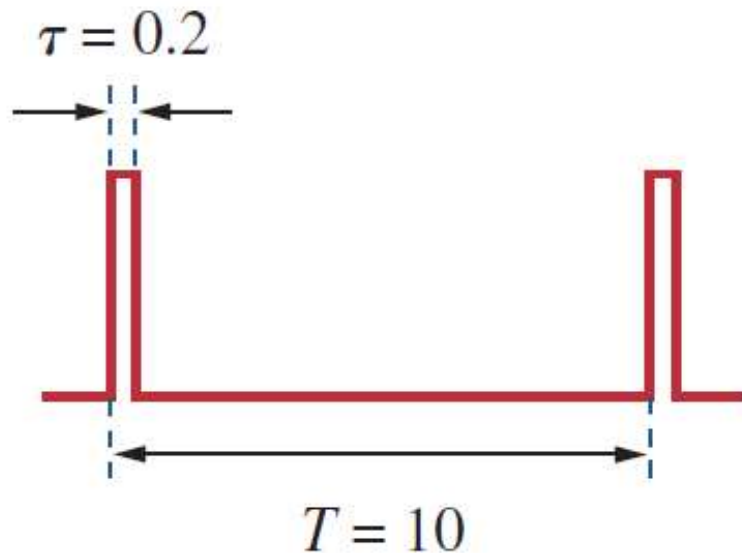
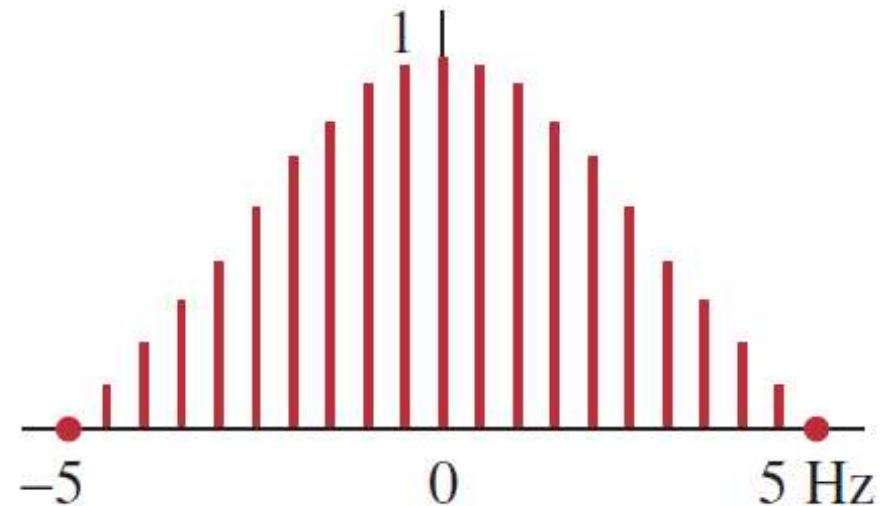
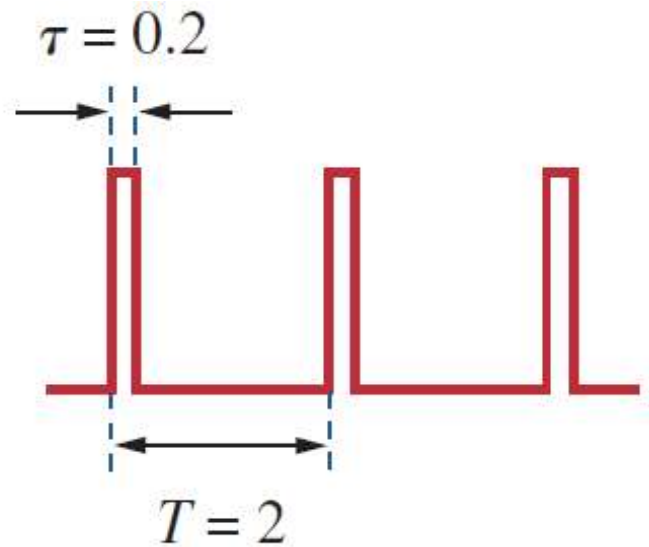
Fourier Transforms

- Observe the spectrum of signal when $T = 1$,
 $A = 10$ and $\tau = 0.2s$



Fourier Transforms

- Observe the spectrum of signal when $T = 2$ & 10 , $A = 10$ and $\tau = 0.2$ s, respectively,



Fourier Transforms

- **The general shape remains same**
- **The frequency at which the envelope first becomes zero remains the same**
- **The amplitude of the spectrum and the spacing between adjacent components both decrease, while the number of harmonics increases**
- **Over a range of frequencies, the sum of the amplitudes of the harmonics remains almost constant**

Fourier Transforms

To further understand the connection between *non-periodic function* and its *periodic counterpart*, let's consider exponential form Fourier series,

$$f(t) = \sum_{n=-\infty}^{\infty} c_n e^{jn\omega_0 t} \quad \text{where,} \quad c_n = \frac{1}{T} \int_{-T/2}^{T/2} f(t) e^{-jn\omega_0 t} dt$$

The fundamental frequency will be, $\omega_0 = \frac{2\pi}{T}$

and the spacing between adjacent harmonics is,

$$\Delta\omega = (n + 1)\omega_0 - n\omega_0 = \omega_0 = \frac{2\pi}{T}$$

Fourier Transforms

Substituting the value of C_n into $f(t)$ and solving summation gives,

$$f(t) = \frac{1}{2\pi} \sum_{n=-\infty}^{\infty} \left[\int_{-T/2}^{T/2} f(t) e^{-jn\omega_0 t} dt \right] \Delta\omega e^{jn\omega_0 t}$$

as $T \rightarrow \infty$

$$\sum_{n=-\infty}^{\infty} \Rightarrow \int_{-\infty}^{\infty}$$

$$\Delta\omega \Rightarrow d\omega$$

$$n\omega_0 \Rightarrow \omega$$

Fourier Transforms

Putting the extensions, the equation becomes,

$$f(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} \left[\int_{-\infty}^{\infty} f(t) e^{-j\omega t} dt \right] e^{j\omega t} d\omega$$

The term in the brackets is known as the *Fourier transform* of $f(t)$ and is represented by $F(\omega)$

$$F(\omega) = \mathcal{F}[f(t)] = \int_{-\infty}^{\infty} f(t) e^{-j\omega t} dt$$

\mathcal{F} is the Fourier operator

Fourier Transforms

- The *Fourier transform* (FT) is an integral transformation of $f(t)$ from the time domain to the frequency domain
- The Fourier transform can also be represented in terms of $F(\omega)$, called *Inverse Fourier transform* (IFT)

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

Transform Pair: $f(t) \Leftrightarrow F(\omega)$

Fourier Transforms

- The *Fourier transform* $F(\omega)$ exist when the Fourier Integral has convergence
- A sufficient but not necessary condition that $f(t)$ has a Fourier transform is that it be completely integrable in the sense that

$$\int_{-\infty}^{\infty} |f(t)| dt < \infty$$

- We will use s instead of $j\omega$ to avoid complexity and replace it after algebraic evaluation

Fourier Transforms – Example 6

Question: Find the Fourier Transform of the following function

$$(a) = \delta(t - t_0)$$

$$(b) = e^{j\omega_o t}$$

$$(c) = \cos(\omega_o t)$$

Fourier Transforms – Example 6

Solution: (a) For the Fourier Transform of the impulse function, we will apply sifting property,

$$F(\omega) = \mathcal{F}[\delta(t - t_0)] = \int_{-\infty}^{\infty} \delta(t - t_0) e^{-j\omega t} dt = e^{-j\omega t_0}$$

For the special case when $t = t_0$

$$\mathcal{F}[\delta(t)] = 1$$

- This shows that the *magnitude* of the spectrum of the impulse function is constant
- All frequencies are equally represented in the impulse function

Fourier Transforms – Example 6

Solution: (b)

Let, $F(\omega) = \delta(\omega - \omega_0)$

then we can write $f(t)$

$$f(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} \delta(\omega - \omega_0) e^{j\omega t} d\omega$$

Using sifting property of impulse function

$$f(t) = \frac{1}{2\pi} e^{j\omega_0 t}$$

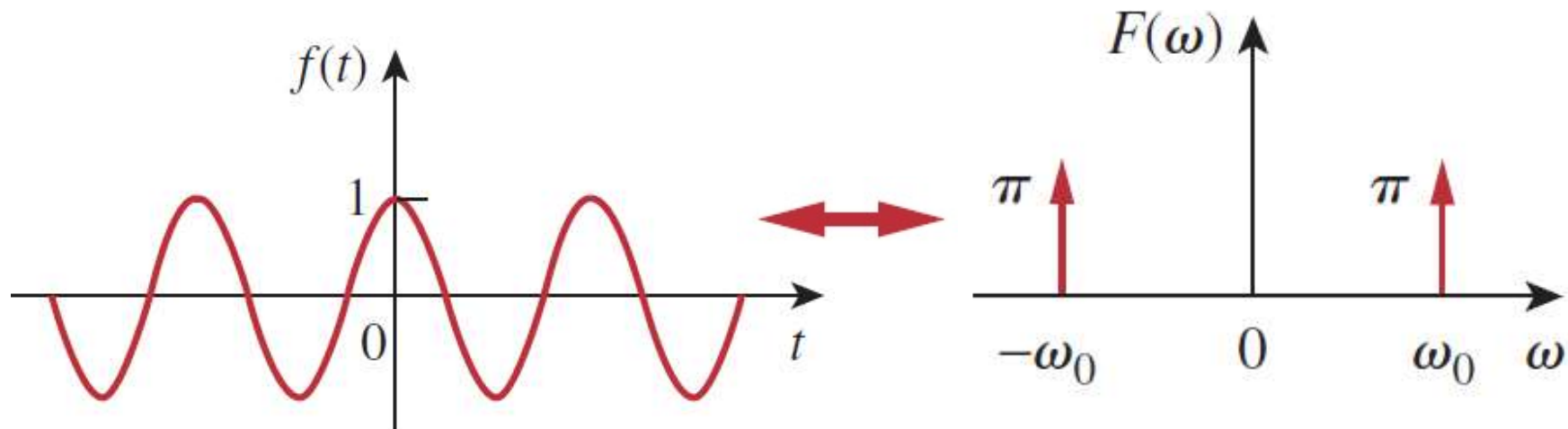
As $f(t)$ and $F(\omega)$ are Fourier transform pair so as $e^{j\omega_0 t}$ and $2\pi\delta(\omega - \omega_0)$

$$\mathcal{F}[e^{j\omega_0 t}] = 2\pi\delta(\omega - \omega_0)$$

Fourier Transforms – Example 6

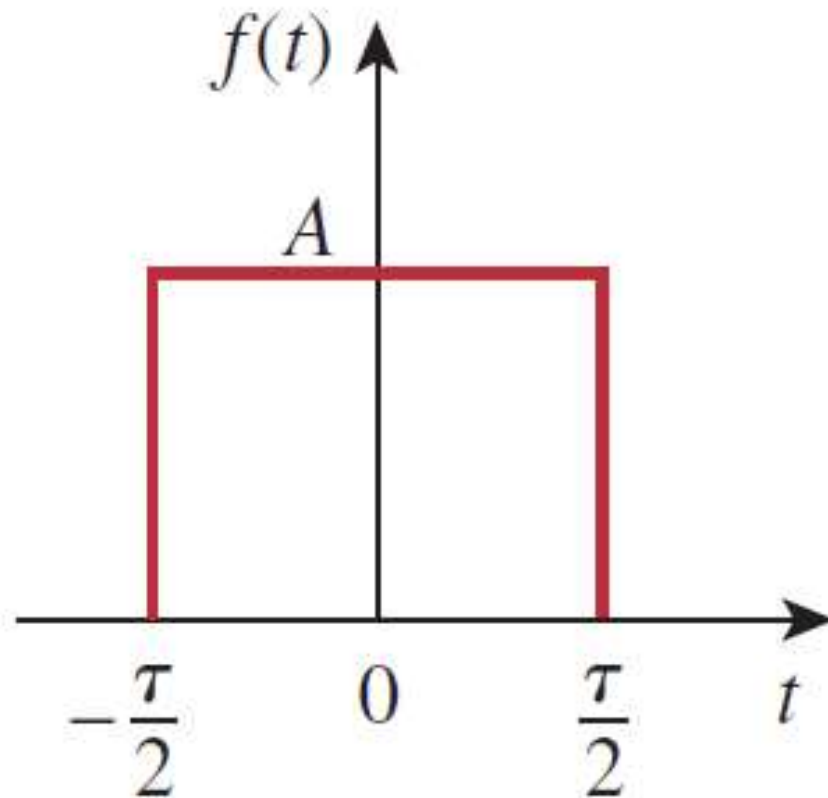
Solution: (c) By using results of (a) and (b), we can write,

$$\begin{aligned}\mathcal{F}[\cos \omega_0 t] &= \mathcal{F}\left[\frac{e^{j\omega_0 t} + e^{-j\omega_0 t}}{2}\right] \\&= \frac{1}{2}\mathcal{F}[e^{j\omega_0 t}] + \frac{1}{2}\mathcal{F}[e^{-j\omega_0 t}] \\&= \pi \delta(\omega - \omega_0) + \pi \delta(\omega + \omega_0)\end{aligned}$$



Fourier Transforms – Example 7

Question: Derive the Fourier transform of a single rectangular pulse (gate function) of width τ and height A ?



Fourier Transforms – Example 7

Solution:

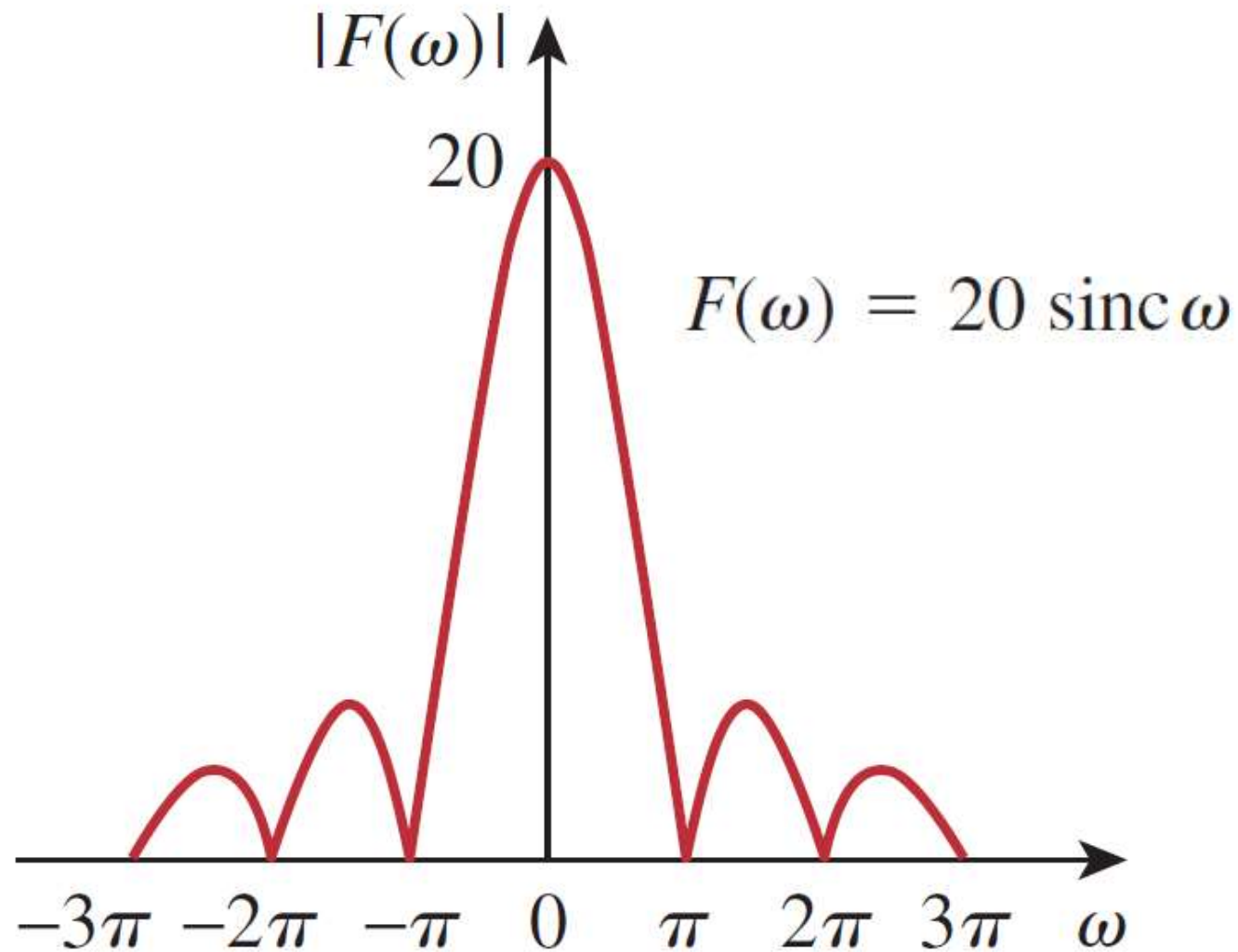
$$\begin{aligned} F(\omega) &= \int_{-\tau/2}^{\tau/2} A e^{-j\omega t} dt = -\frac{A}{j\omega} e^{-j\omega t} \Big|_{-\tau/2}^{\tau/2} \\ &= \frac{2A}{\omega} \left(\frac{e^{j\omega\tau/2} - e^{-j\omega\tau/2}}{2j} \right) \\ &= A\tau \frac{\sin \omega\tau/2}{\omega\tau/2} = A\tau \operatorname{sinc} \frac{\omega\tau}{2} \end{aligned}$$

If we take $A = 10$, and $\tau = 2$ s, then the solution will be a sinc function

$$F(\omega) = 20 \operatorname{sinc} \omega$$

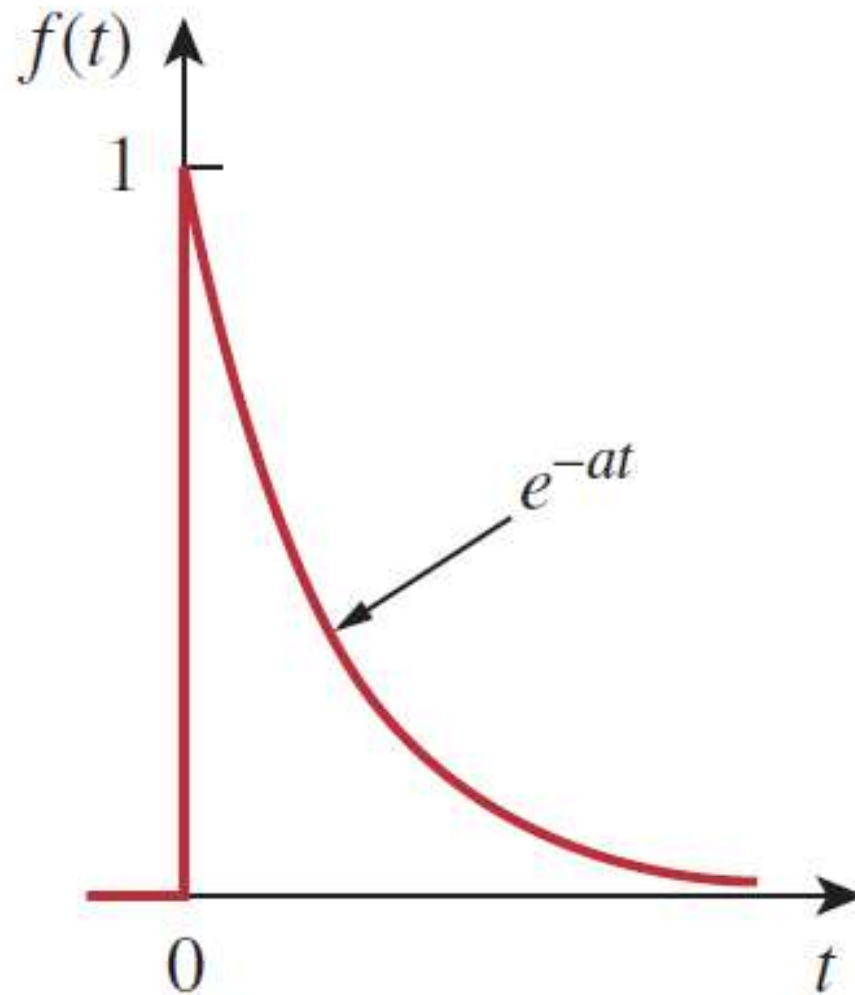
Fourier Transforms – Example 7

Amplitude spectrum



Fourier Transforms – Example 7

Question: Derive the Fourier transform of a “switched ON” exponential function shown below?



Fourier Transforms – Example 7

Solution:

$$f(t) = e^{-at}u(t) = \begin{cases} e^{-at}, & t > 0 \\ 0, & t < 0 \end{cases}$$

Hence,

$$\begin{aligned} F(\omega) &= \int_{-\infty}^{\infty} f(t)e^{-j\omega t} dt \\ &= \int_0^{\infty} e^{-at} e^{-j\omega t} dt = \int_0^{\infty} e^{-(a+j\omega)t} dt \\ &= \frac{-1}{a + j\omega} e^{-(a+j\omega)t} \Bigg|_0^{\infty} = \frac{1}{a + j\omega} \end{aligned}$$

Objectives

- Average and rms signal power
- Parseval's theorem
- Total harmonic distortion
- Fourier Transforms
- **Properties of Fourier transform**
- Circuit interpretation for Fourier transforms
- Signal Energy and Spectrum
- Signal Bandwidth

Properties of Fourier Transforms

Linearity:

$$\mathcal{F}[a_1 f_1(t) + a_2 f_2(t)] = a_1 F_1(\omega) + a_2 F_2(\omega)$$

a_1 and a_2 are two arbitrary constants

Examp^r:

$$\begin{aligned} F[\sin \omega_0 t] &= \frac{1}{2j} [\mathcal{F}(e^{j\omega_0 t}) - \mathcal{F}(e^{-j\omega_0 t})] \\ &= \frac{\pi}{j} [\delta(\omega - \omega_0) - \delta(\omega + \omega_0)] \\ &= j\pi [\delta(\omega + \omega_0) - \delta(\omega - \omega_0)] \end{aligned}$$

Properties of Fourier Transforms

Time scaling: If $F(\omega) = \mathcal{F}(f(t))$, then

$$\mathcal{F}[f(at)] = \frac{1}{|a|} F\left(\frac{\omega}{a}\right)$$

a is arbitrary constants

Properties of Fourier Transforms

Time shifting: If $F(\omega) = \mathcal{F}(f(t))$, then

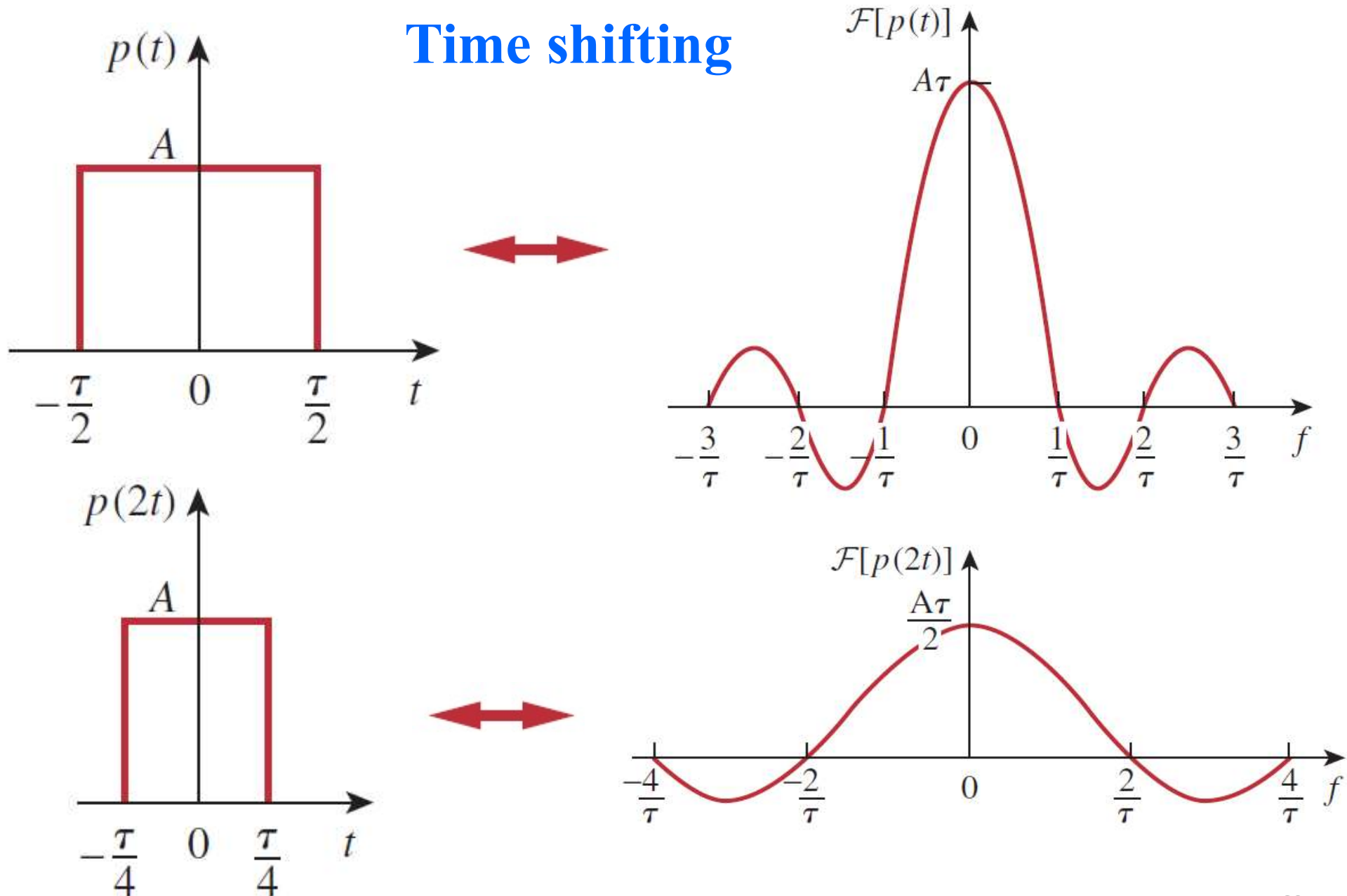
$$\mathcal{F}[f(t - t_0)] = e^{-j\omega t_0} F(\omega)$$

t_0 is arbitrary constants used for delay

- A delay in the time domain corresponds to a phase shift in the frequency domain

Properties of Fourier Transforms

Time shifting



Properties of Fourier Transforms

Frequency shifting (Amplitude modulation):

If $F(\omega) = \mathcal{F}(f(t))$, then

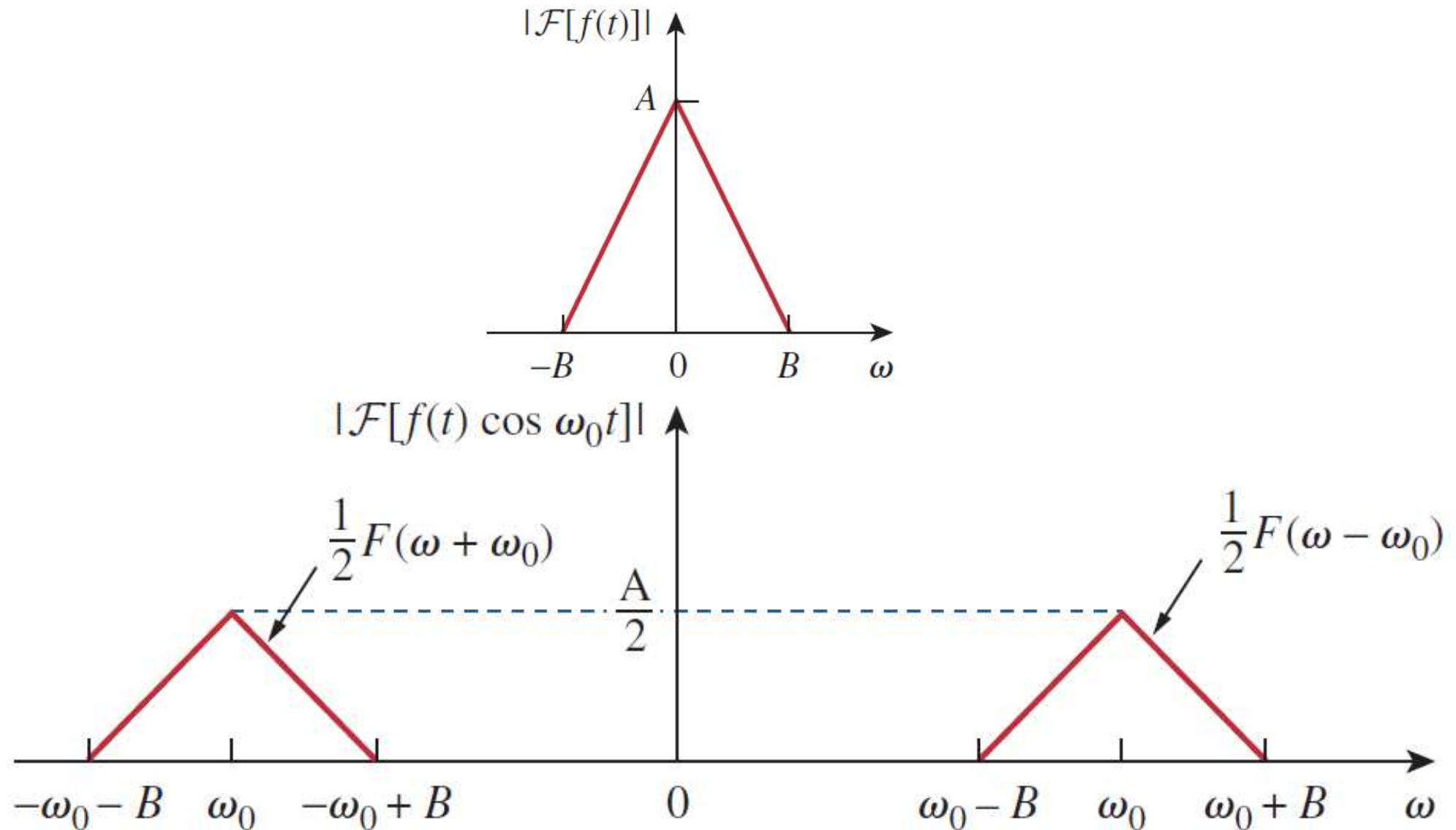
$$\mathcal{F}[f(t)e^{j\omega_0 t}] = F(\omega - \omega_0)$$

ω_0 is arbitrary constants used for delay

- A frequency shift in the frequency domain adds a phase shift to the time function

Properties of Fourier Transforms

Frequency shifting (Amplitude modulation)



Properties of Fourier Transforms

Time differentiation:

If $F(\omega) = \mathcal{F}(f(t))$, then

$$\mathcal{F}[f'(t)] = j\omega F(\omega)$$

- The transform of the derivative of $f(t)$ is obtained by multiplying the transform of $f(t)$ by $j\omega$
- For higher order differentiation,

$$\mathcal{F}[f^{(n)}(t)] = (j\omega)^n F(\omega)$$

Properties of Fourier Transforms

Time integration: If $F(\omega) = \mathcal{F}(f(t))$, then

$$\mathcal{F}\left[\int_{-\infty}^t f(t) dt\right] = \frac{F(\omega)}{j\omega} + \pi F(0) \delta(\omega)$$

- The transform of the integral of $f(t)$ is obtained by dividing the transform of $f(t)$ by $j\omega$ and *adding* the result to the impulse term that reflects the **dc component** $F(0)$

Properties of Fourier Transforms

Time Reversal: If $F(\omega) = \mathcal{F}(f(t))$, then

$$\mathcal{F}[f(-t)] = F(-\omega) = F^*(\omega)$$

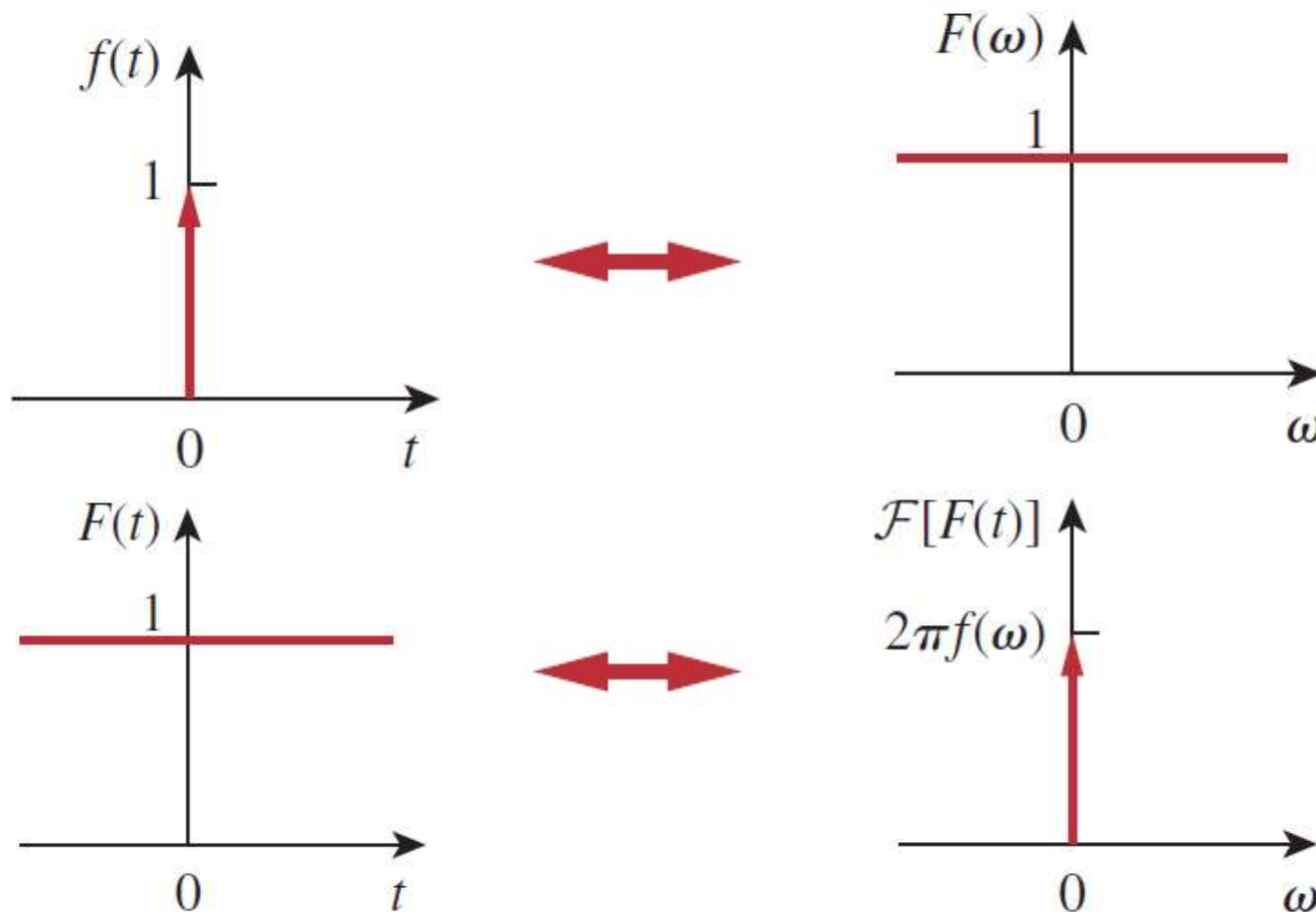
*** is complex conjugate**

- Reversing $f(t)$ about the time axis reverses $F(\omega)$ about the frequency axis

Properties of Fourier Transforms

Duality: If $F(\omega) = \mathcal{F}(f(t))$, then

$$\mathcal{F}[f(t)] = F(\omega) \quad \Rightarrow \quad \mathcal{F}[F(t)] = 2\pi f(-\omega)$$



Properties of Fourier Transforms

Convolution:

If $x(t)$ is the input excitation to a circuit with an impulse function of $h(t)$, then the output response $y(t)$ is given by the convolution integral

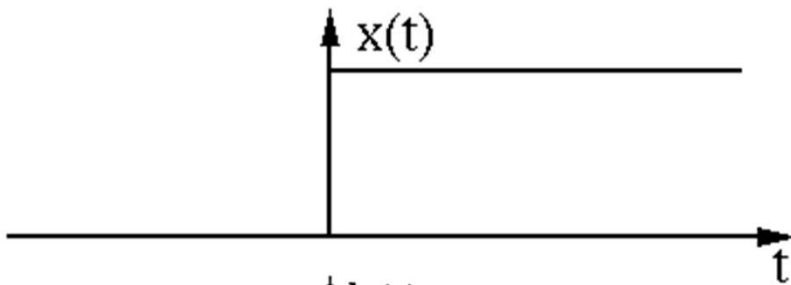
$$y(t) = h(t) * x(t) = \int_{-\infty}^{\infty} h(\lambda) x(t - \lambda) d\lambda$$

if $X(\omega)$, $H(\omega)$ and $Y(\omega)$ are Fourier transform of $x(t)$, $h(t)$ and $y(t)$, respectively, then

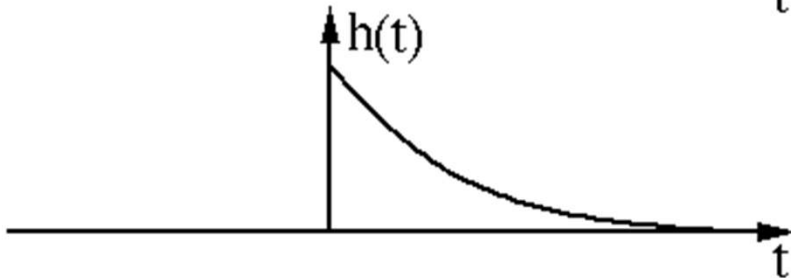
$$Y(\omega) = \mathcal{F}[h(t) * x(t)] = H(\omega)X(\omega)$$

Convolution –Graphical illustration

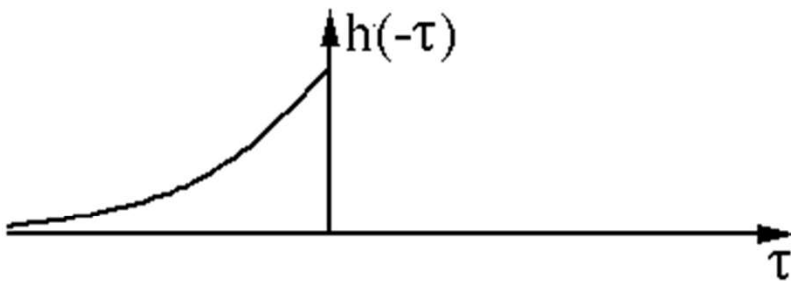
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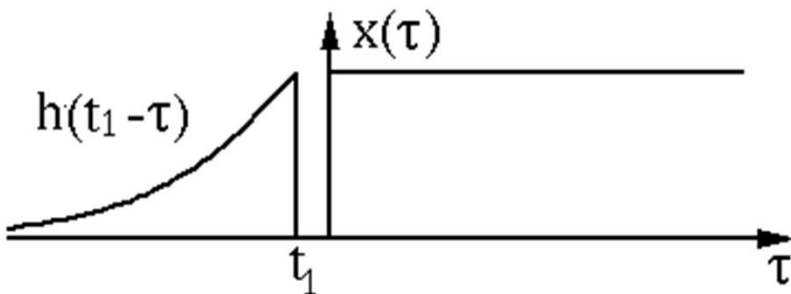
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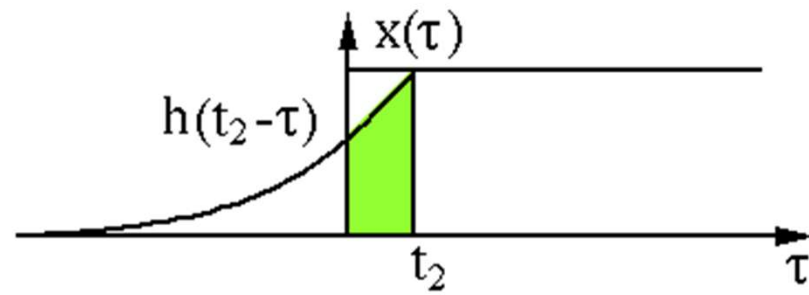
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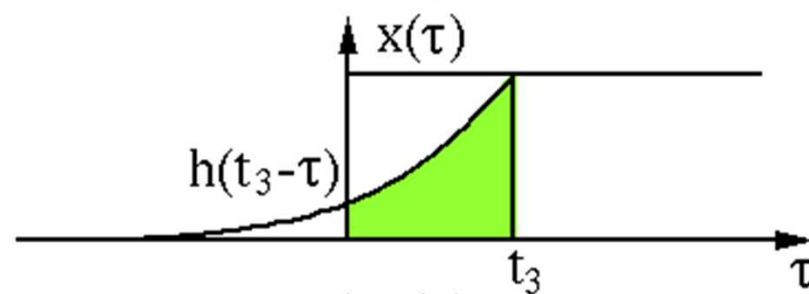
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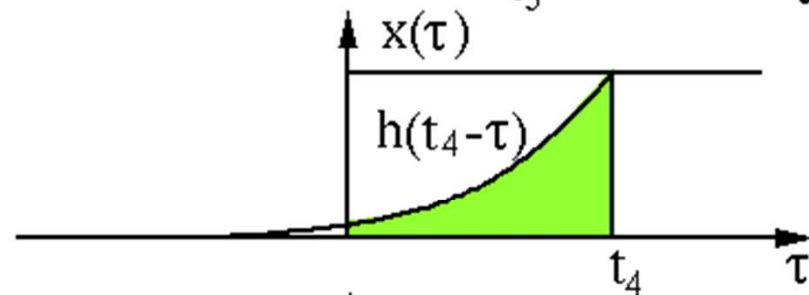
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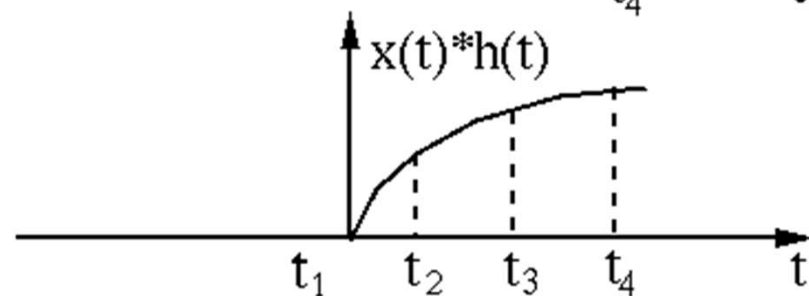
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7



8



Fourier Transforms – Summary

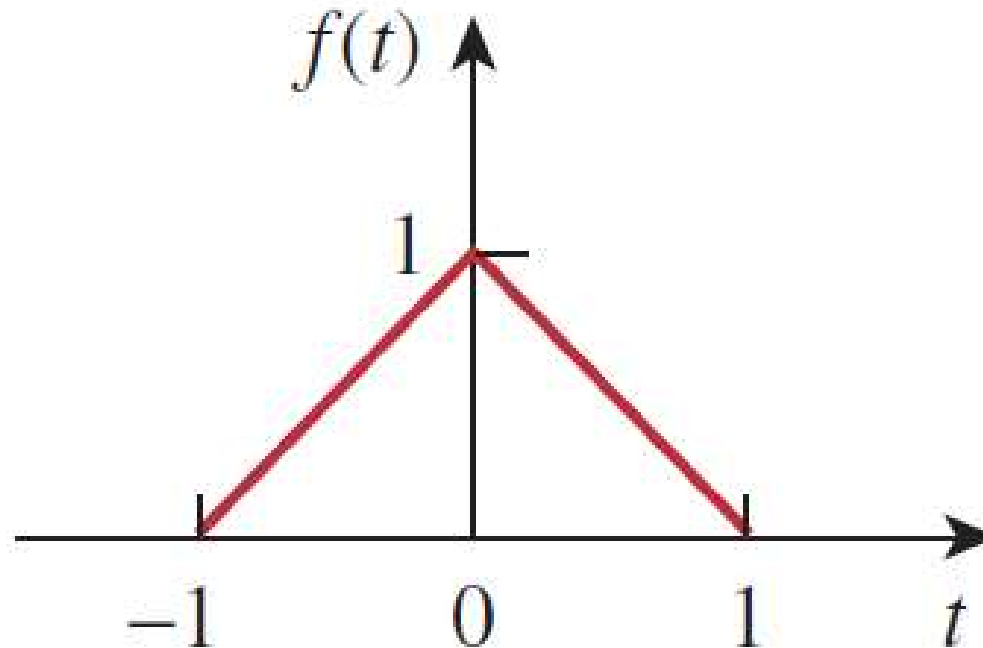
Property	$f(t)$	$F(\omega)$
Linearity	$a_1 f_1(t) + a_2 f_2(t)$	$a_1 F_1(\omega) + a_2 F_2(\omega)$
Scaling	$f(at)$	$\frac{1}{ a } F\left(\frac{\omega}{a}\right)$
Time shift	$f(t - a)$	$e^{-j\omega a} F(\omega)$
Frequency shift	$e^{j\omega_0 t} f(t)$	$F(\omega - \omega_0)$
Modulation	$\cos(\omega_0 t) f(t)$	$\frac{1}{2} [F(\omega + \omega_0) + F(\omega - \omega_0)]$
Time differentiation	$\frac{df}{dt}$	$j\omega F(\omega)$
	$\frac{d^n f}{dt^n}$	$(j\omega)^n F(\omega)$

Fourier Transforms – Summary

Property	$f(t)$	$F(\omega)$
Time integration	$\int_{-\infty}^t f(t) dt$	$\frac{F(\omega)}{j\omega} + \pi F(0) \delta(\omega)$
Frequency differentiation	$t^n f(t)$	$(j)^n \frac{d^n}{d\omega^n} F(\omega)$
Reversal	$f(-t)$	$F(-\omega)$ or $F^*(\omega)$
Duality	$F(t)$	$2\pi f(-\omega)$
Convolution in t	$f_1(t) * f_2(t)$	$F_1(\omega) F_2(\omega)$
Convolution in ω	$f_1(t) f_2(t)$	$\frac{1}{2\pi} F_1(\omega) * F_2(\omega)$

Fourier Transforms – Example 8

Question: Find the Fourier transform of the following signal?

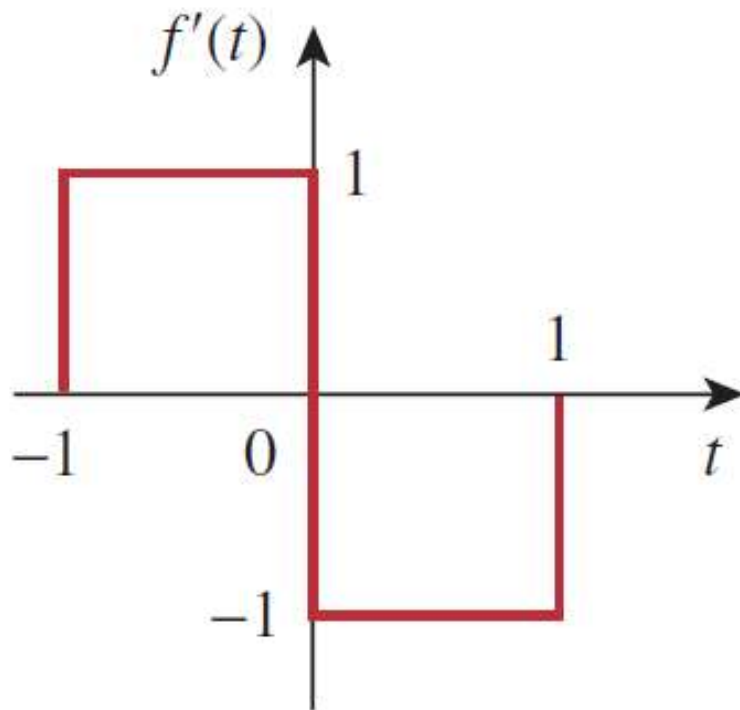


Fourier Transforms – Example 8

Solution: Applying derivative property will lead to the easy path towards solution, The signal can be written as,

$$f(t) = \begin{cases} 1 + t, & -1 < t < 0 \\ 1 - t, & 0 < t < 1 \end{cases}$$

It's first derivative will result into,

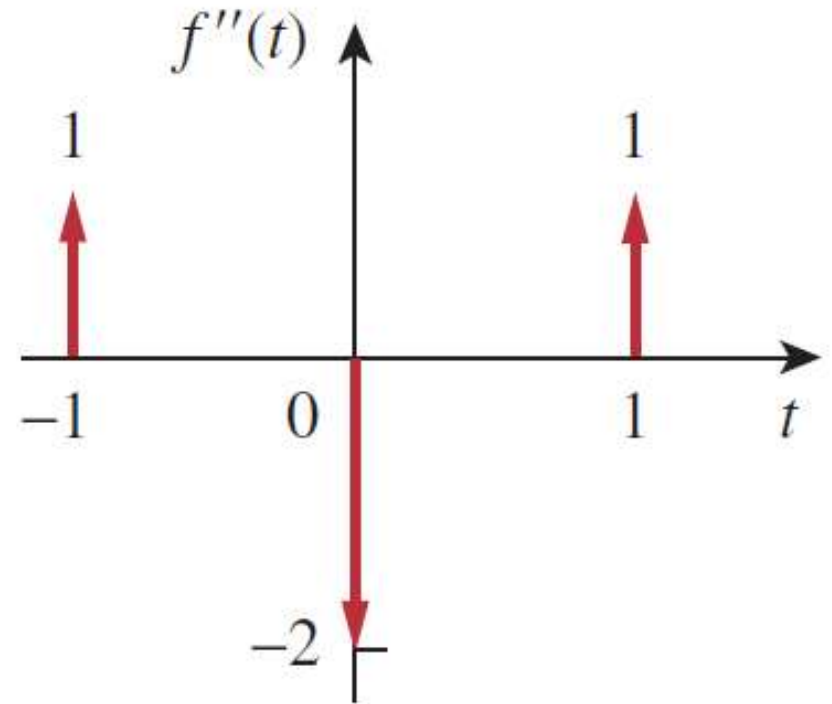


$$f'(t) = \begin{cases} 1, & -1 < t < 0 \\ -1, & 0 < t < 1 \end{cases}$$

Fourier Transforms – Example 8

It's second derivative

$$f''(t) = \delta(t + 1) - 2\delta(t) + \delta(t - 1)$$



taking Fourier transform
of both sides,

$$(j\omega)^2 F(\omega) = e^{j\omega} - 2 + e^{-j\omega} = -2 + 2 \cos \omega$$

$$F(\omega) = \frac{2(1 - \cos \omega)}{\omega^2}$$

Fourier Transforms – Example 9

Question: Obtain the inverse Fourier transform of following function?

$$(a) F(\omega) = \frac{10j\omega + 4}{(j\omega)^2 + 6j\omega + 8}$$

$$(b) G(\omega) = \frac{\omega^2 + 21}{\omega^2 + 9}$$

Fourier Transforms – Example 9

Solution: (a)

To avoid algebraic complexity, we use s instead of $j\omega$ for a moment and replace later on, Using partial fraction expansion

$$\begin{aligned} F(s) &= \frac{10s + 4}{s^2 + 6s + 8} \\ &= \frac{10s + 4}{(s + 4)(s + 2)} \\ &= \frac{A}{s + 4} + \frac{B}{s + 2} \end{aligned}$$

Fourier Transforms – Example 9

$$A = (s + 4)F(s) \big|_{s=-4} = \frac{10s + 4}{(s + 2)} \bigg|_{s=-4} = \frac{-36}{-2} = 18$$

$$B = (s + 2)F(s) \big|_{s=-2} = \frac{10s + 4}{(s + 4)} \bigg|_{s=-2} = \frac{-16}{2} = -8$$

Substituting the values of A , B and replacing **s** with **$j\omega$**

$$F(j\omega) = \frac{18}{j\omega + 4} + \frac{-8}{j\omega + 2}$$

using transform pair,

$$f(t) = (18e^{-4t} - 8e^{-2t})u(t)$$

Fourier Transforms – Example 9

Solution: (b)

Applying simplification for $G(\omega)$,

$$G(\omega) = \frac{\omega^2 + 21}{\omega^2 + 9} = 1 + \frac{12}{\omega^2 + 9}$$

using transform pair,

$$g(t) = \delta(t) + 2e^{-3|t|}$$

Objectives

- Average and rms signal power
- Parseval's theorem
- Total harmonic distortion
- Fourier Transforms
- Properties of Fourier transform
- **Circuit interpretation for Fourier transforms**
- Signal Energy and Spectrum
- Signal Bandwidth

Circuit solution with Fourier Transform

- **The Fourier transform generalizes the phasor technique to non-periodic functions**
- **We apply Fourier transforms to circuits with non-sinusoidal excitations in exactly the same way we apply phasor techniques to circuits with sinusoidal excitations**
- **Ohm's law is still valid**

$$V(\omega) = Z(\omega)I(\omega)$$

- **Fourier transform cannot solve initial conditions**

Circuit solution with Fourier Transform

- We transform the functions for the circuit elements into the *frequency domain* and take the *Fourier transforms of the excitations*, then we can apply
 - ✓ Voltage/current division
 - ✓ Source transformation,
 - ✓ Mesh analysis
 - ✓ Node analysis
 - ✓ Thevenin's theorem
 - ✓ Norton's theorem
- We take the inverse Fourier transform to obtain the response in the time domain

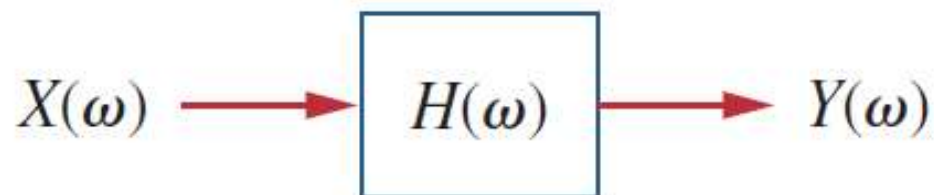
Circuit solution with Fourier Transform

***RLC* components
Transforms**



R	\Rightarrow	R
L	\Rightarrow	$j\omega L$
C	\Rightarrow	$\frac{1}{j\omega C}$

Transfer function

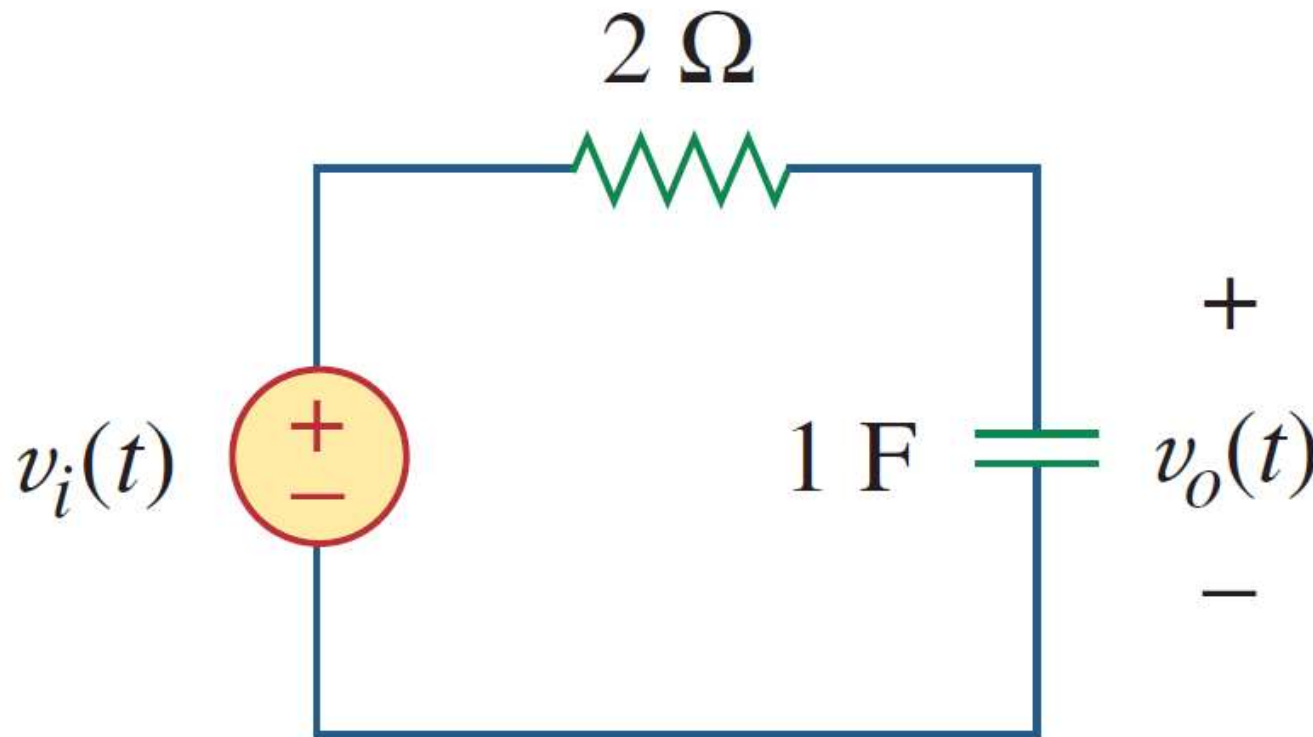


$$H(\omega) = \frac{Y(\omega)}{X(\omega)}$$

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

Fourier Transform – Example 10

Question: Find $v_o(t)$ if the $v_i(t) = 2e^{-3t}u(t)$?



Fourier Transform – Example 10

Solution: The Fourier transform of the input voltage $v_i(t) = 2e^{-3t}u(t)$ is,

$$V_i(\omega) = \frac{2}{3 + j\omega}$$

and the transfer function by using voltage division is,

$$H(\omega) = \frac{V_o(\omega)}{V_i(\omega)} = \frac{1/j\omega}{2 + 1/j\omega} = \frac{1}{1 + j2\omega}$$

$$V_o(\omega) = V_i(\omega)H(\omega) = \frac{2}{(3 + j\omega)(1 + j2\omega)}$$

Fourier Transform – Example 10

$$V_o(\omega) = \frac{1}{(3 + j\omega)(0.5 + j\omega)}$$

By applying partial fraction,

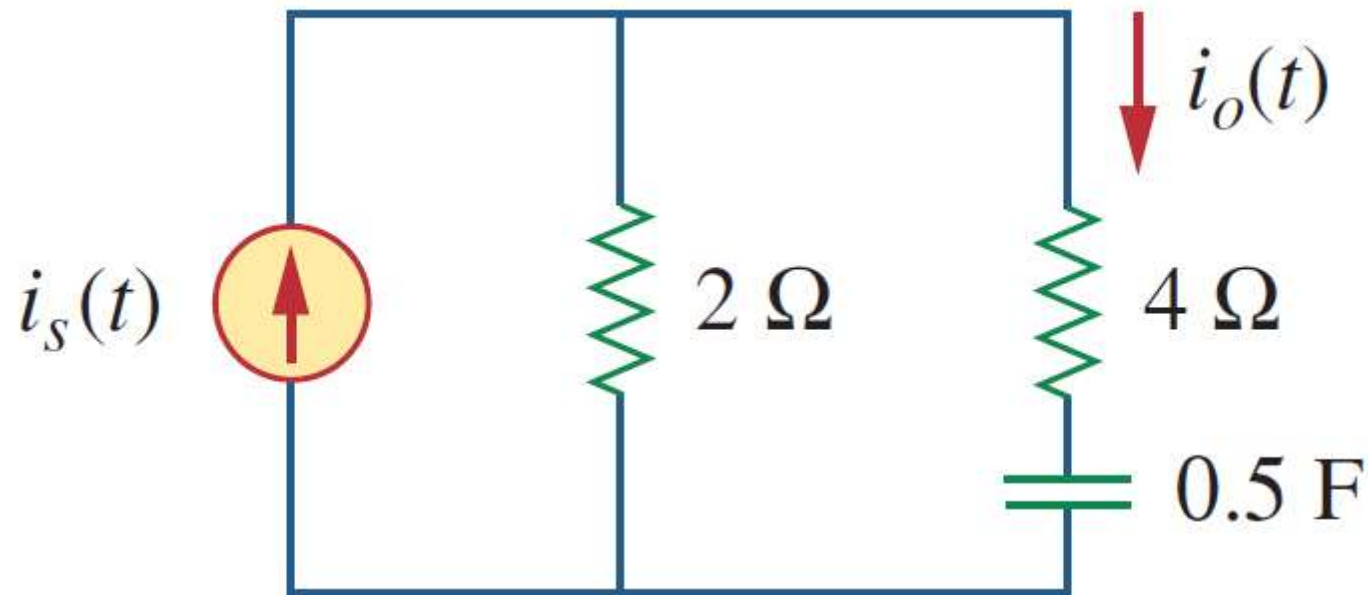
$$V_o(\omega) = \frac{-0.4}{3 + j\omega} + \frac{0.4}{0.5 + j\omega}$$

taking the inverse Fourier transform gives,

$$v_o(t) = 0.4(e^{-0.5t} - e^{-3t})u(t)$$

Fourier Transform – Example 11

Question: Find $i_o(t)$ if the $i_s(t) = 10 \sin(2t)$ A?



Fourier Transform – Example 11

Solution: Using current division rule gives,

$$H(\omega) = \frac{I_o(\omega)}{I_s(\omega)} = \frac{2}{2 + 4 + 2/j\omega} = \frac{j\omega}{1 + j\omega 3}$$

If $i_s(t) = 10 \sin(2t)$ then,

$$I_s(\omega) = j\pi 10[\delta(\omega + 2) - \delta(\omega - 2)]$$

$$I_o(\omega) = H(\omega)I_s(\omega) = \frac{10\pi\omega[\delta(\omega - 2) - \delta(\omega + 2)]}{1 + j\omega 3}$$

Fourier Transform – Example 11

Using inverse Fourier transform pairs,

$$\begin{aligned} i_o(t) &= \mathcal{F}^{-1}[I_o(\omega)] \\ &= \frac{1}{2\pi} \int_{-\infty}^{\infty} \frac{10\pi\omega[\delta(\omega - 2) - \delta(\omega + 2)]}{1 + j\omega 3} e^{j\omega t} d\omega \end{aligned}$$

Using sifting property of impulse function,

or,
$$\delta(\omega - \omega_0)f(\omega) = f(\omega_0)$$

$$\int_{-\infty}^{\infty} \delta(\omega - \omega_0)f(\omega) d\omega = f(\omega_0)$$

Fourier Transform – Example 11

We obtain $i_o(t)$ as,

$$\begin{aligned} i_o(t) &= \frac{10\pi}{2\pi} \left[\frac{2}{1+j6} e^{j2t} - \frac{-2}{1-j6} e^{-j2t} \right] \\ &= 10 \left[\frac{e^{j2t}}{6.082 e^{j80.54^\circ}} + \frac{e^{-j2t}}{6.082 e^{-j80.54^\circ}} \right] \\ &= 1.644 [e^{j(2t-80.54^\circ)} + e^{-j(2t-80.54^\circ)}] \\ &= 3.288 \cos(2t - 80.54^\circ) \text{ A} \end{aligned}$$

Objectives

- Average and rms signal power
- Parseval's theorem
- Total harmonic distortion
- Fourier Transforms
- Properties of Fourier transform
- Circuit interpretation for Fourier transforms
- **Signal Energy and Spectrum**
- Signal Bandwidth

Signal Energy

- The concept of average power cannot be applied on *aperiodic signals*
- The period information required for average power is missing
- The Energy of signal is thus comprising range $[-\infty, \infty]$, given by

$$W = \int_{-\infty}^{\infty} |f(t)|^2 dt$$

$f(t)$ is an aperiodic signal

Signal Energy

- We cannot take Fourier series coefficients due to infinite limits
- It is practical to take FT (Fourier Transform) of the signal
- The Parseval's theorem –Now, Rayleigh's theorem for aperiodic signal will be,

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

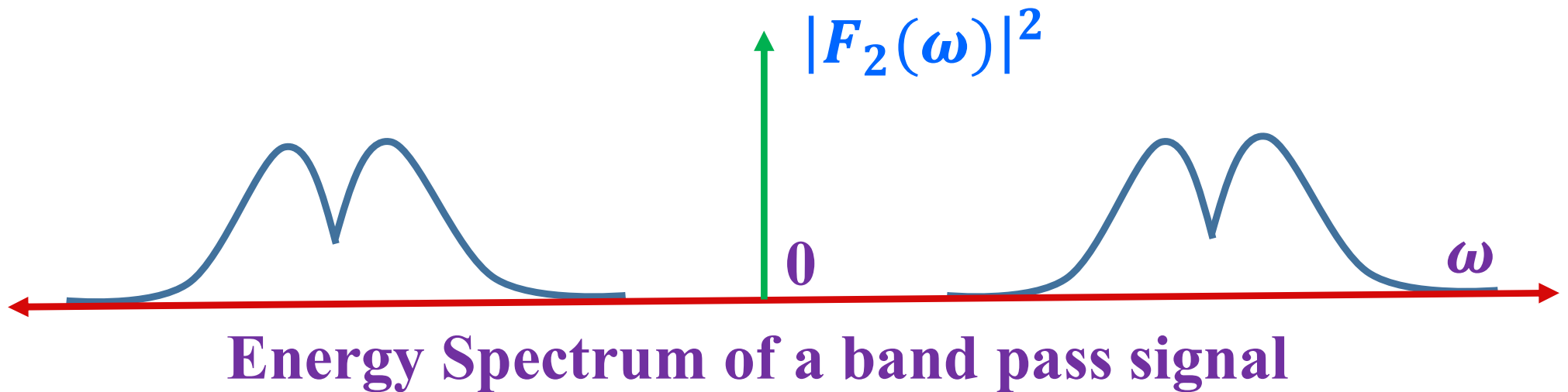
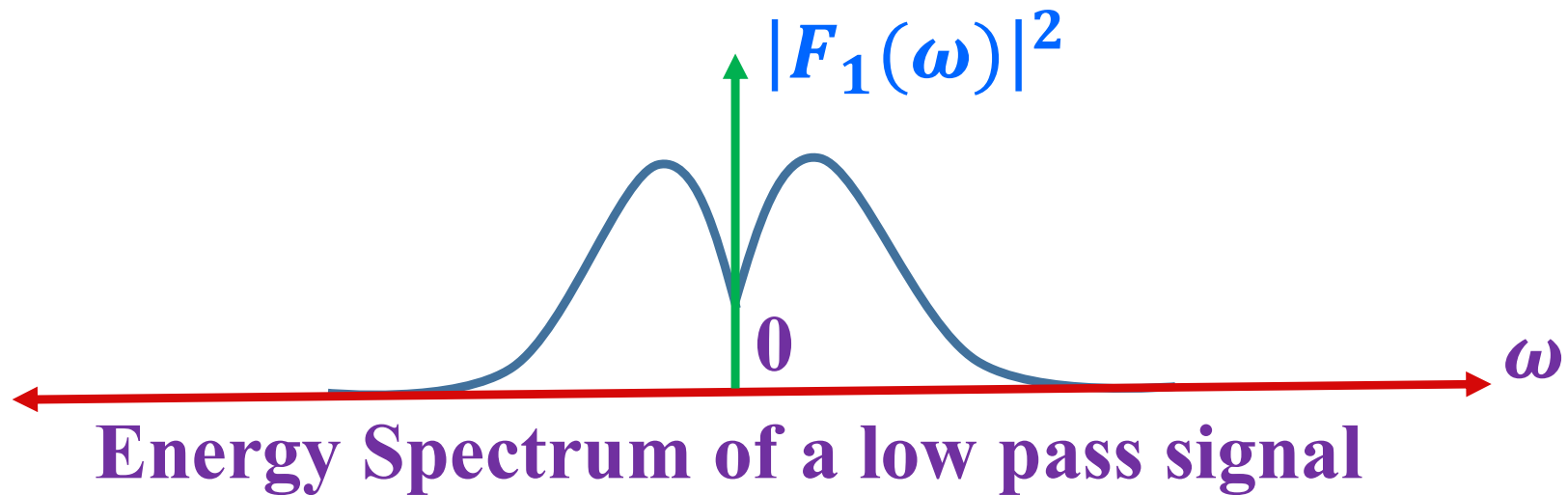
Signal Energy

- Parseval's theorem applies on all finite energy signals
- Any signal having finite period is called energy signal
- Note that;

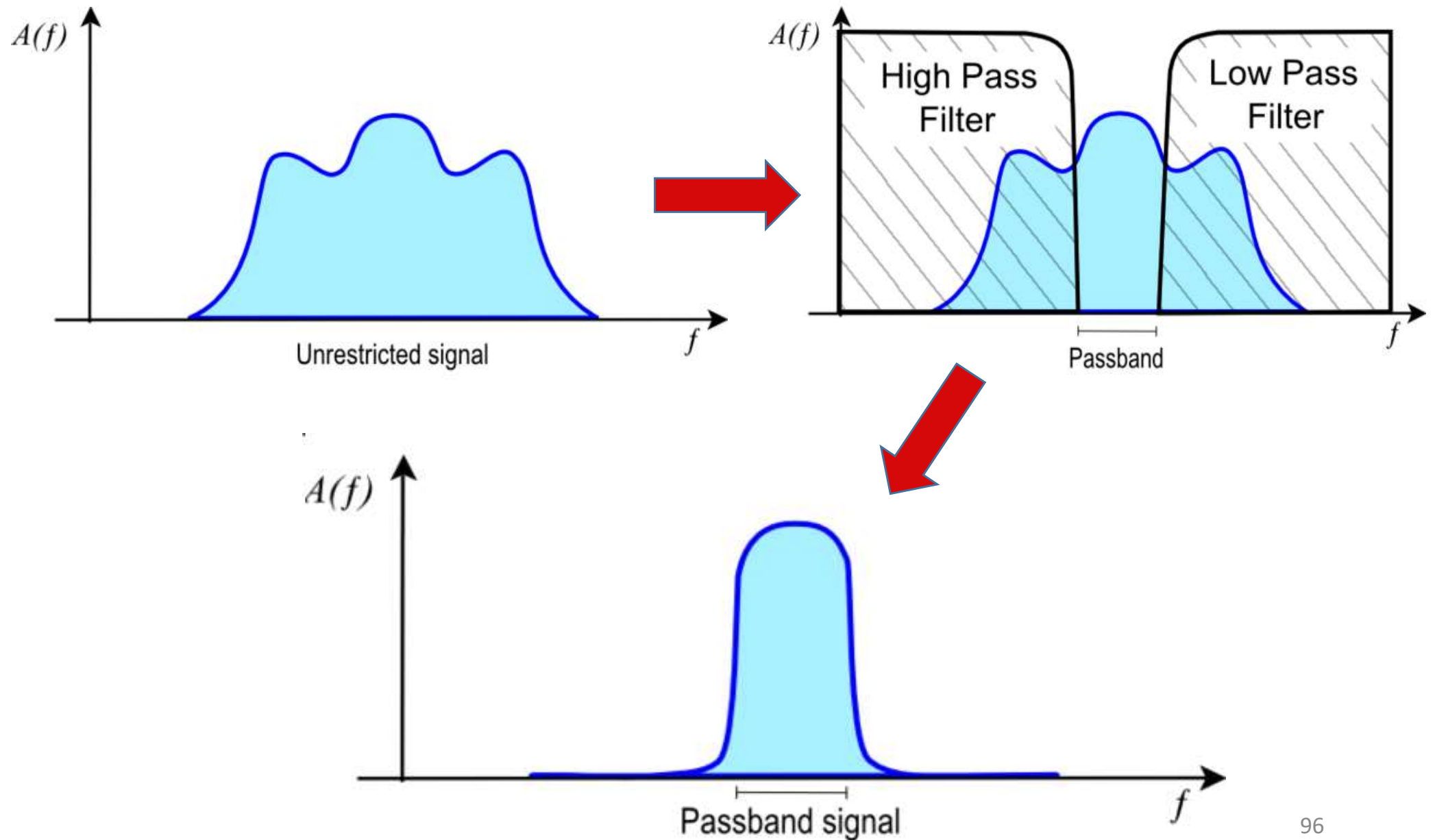
Periodic Signal ➡ $|F_n|^2$ ➡ **Power Spectrum**

Aperiodic Signal ➡ $|F(\omega)|^2$ ➡ **Energy Spectrum**

Signal Spectrum – Example Signals



Signal Energy – PB Example

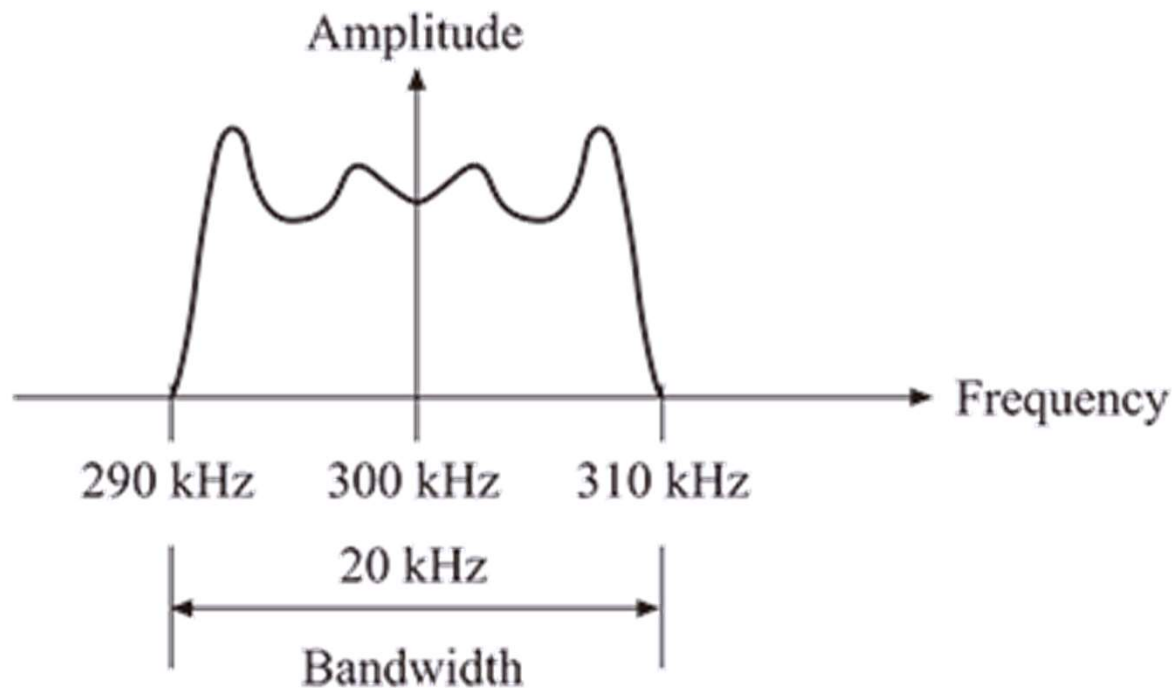


Objectives

- Average and rms signal power
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Signal Bandwidth

- Bandwidth is the **span of frequencies** within the spectrum that is **occupied** by a given signal
- For example: For $BW=20$ KHz



Signal Bandwidth –LP Bandwidth

- Bandwidth for low pass signal corresponds to a positive frequency as

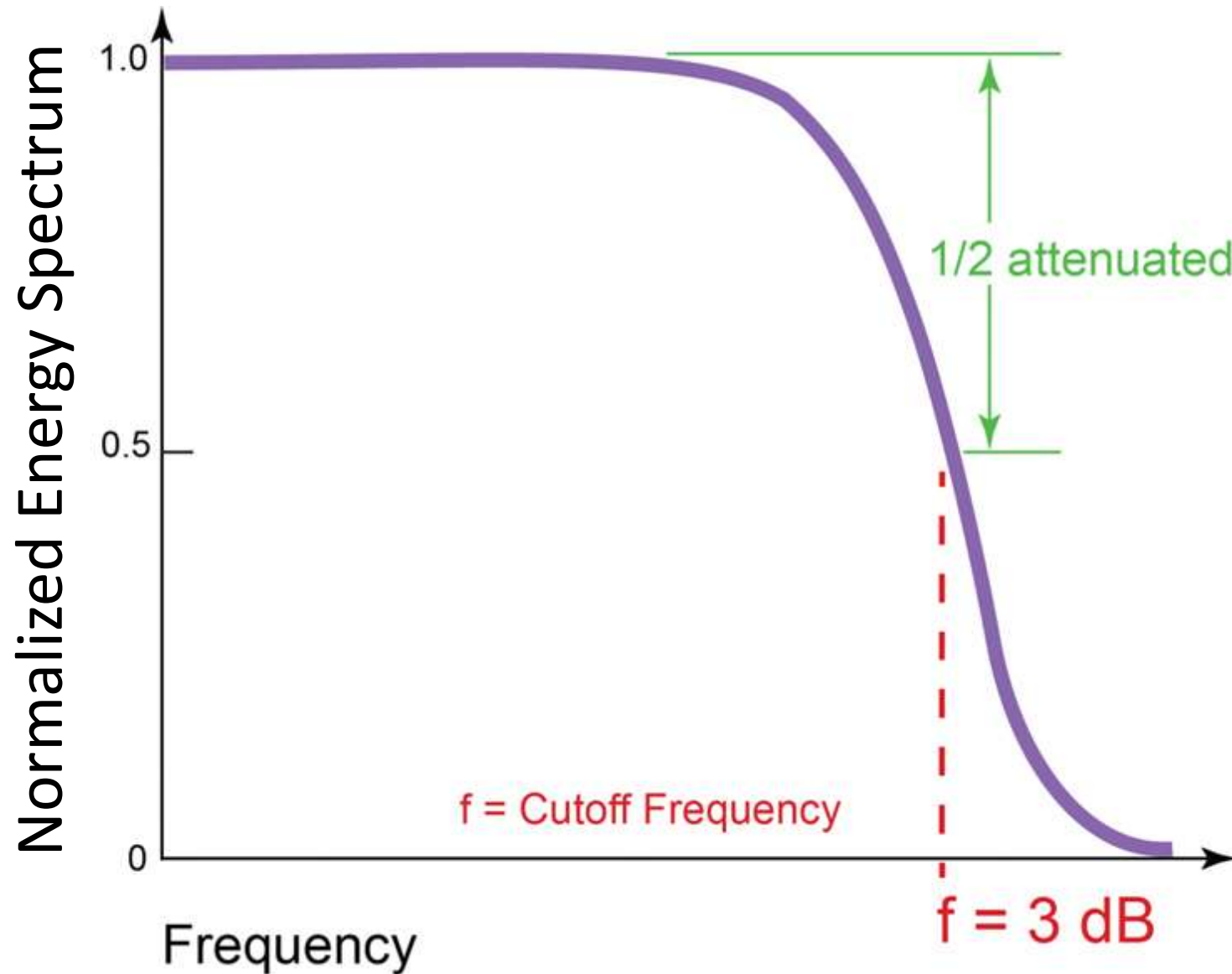
$$\omega = \Omega = 2\pi B$$

- Beyond this, the energy spectrum $|F(\omega)|^2$ is too small to be ignored
- Mostly, we consider *3-dB bandwidth*

$$3\text{-dB bandwidth} = \frac{|F(\Omega)|^2}{|F(0)|^2} = \frac{1}{2}$$

- The energy spectrum falls to one half of the spectral value $|F(0)|^2$ at DC level

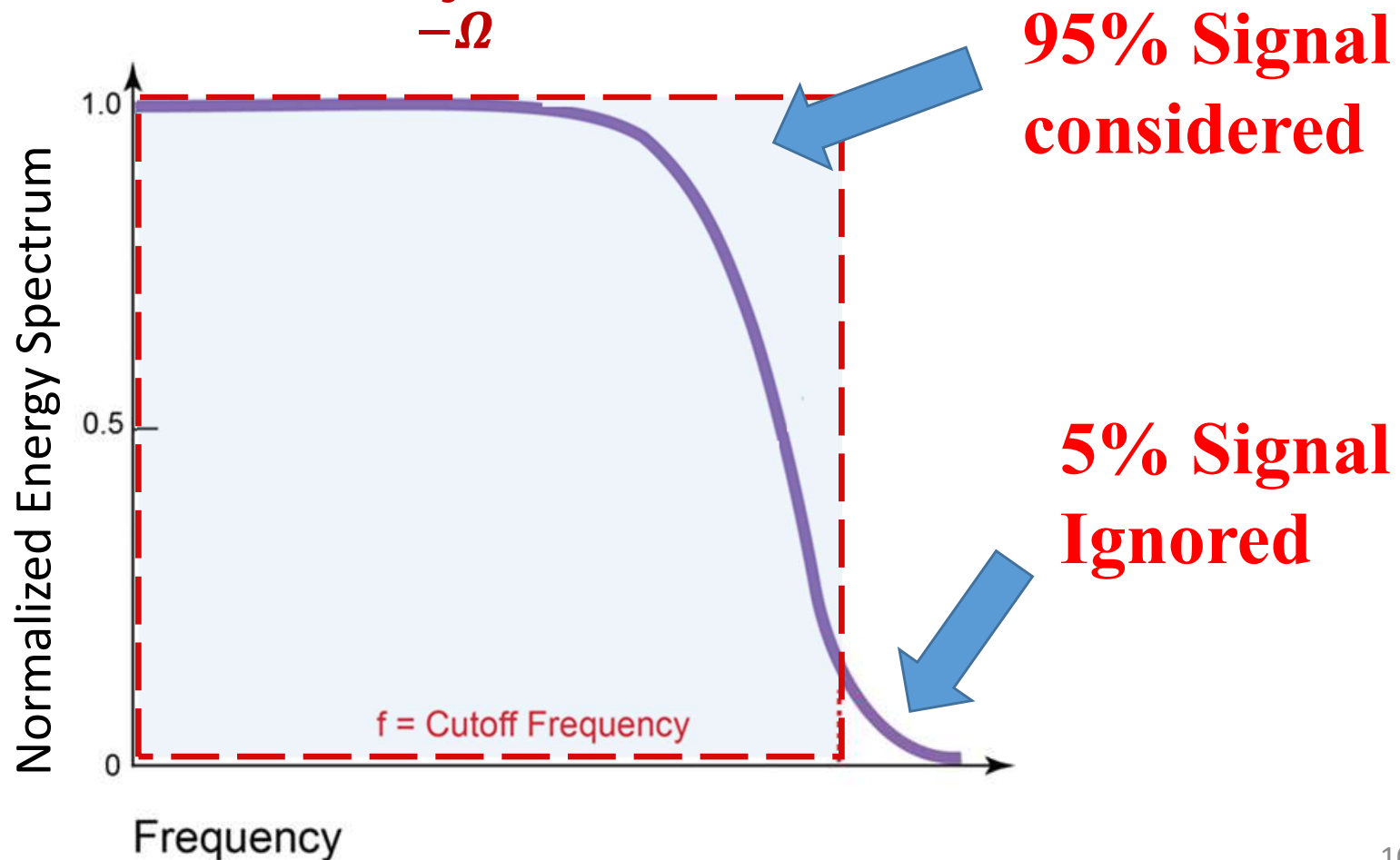
Signal Bandwidth –LP Bandwidth



Signal Bandwidth –LP Bandwidth

- Another definition for $\Omega = 2\pi B$ requires that,

$$\frac{1}{2\pi} \int_{-\Omega}^{\Omega} |F(\omega)|^2 d\omega = rW$$



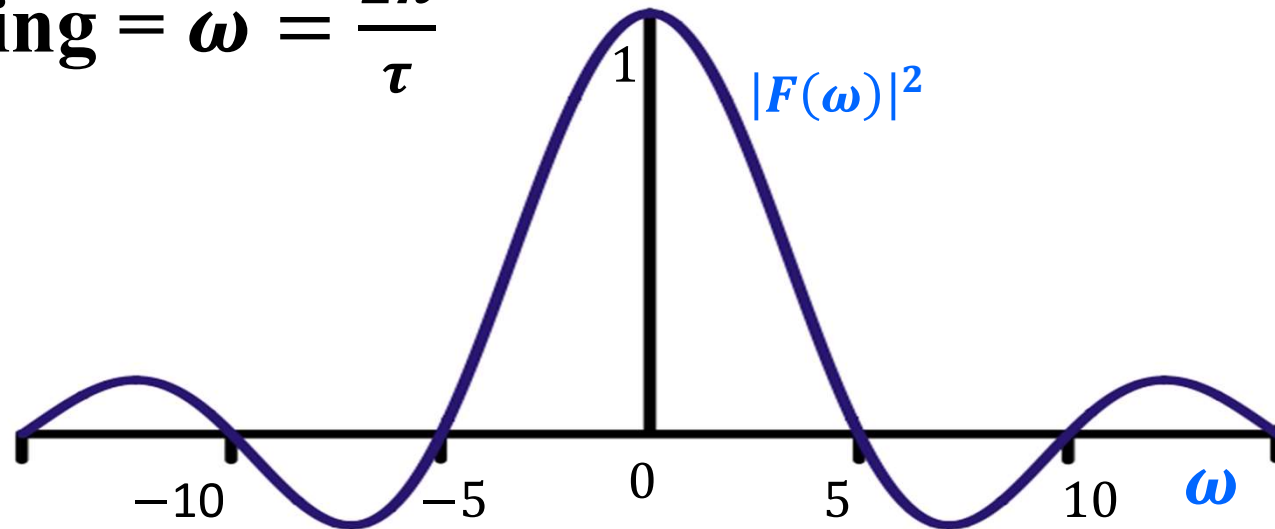
Signal Bandwidth – Example 8

Question: The Fourier transform of the signal,

$$f(t) = \text{rect}\left(\frac{t}{\tau}\right) \quad \text{is} \quad F(\omega) = \tau \cdot \text{sinc}\left(\frac{\omega\tau}{2}\right)$$

So, the corresponding energy signal, $|F(\omega)|^2 = \tau^2 \text{sinc}^2\left(\frac{\omega\tau}{2}\right)$

First zero-crossing = $\omega = \frac{2\pi}{\tau}$



Show that choosing $\Omega = 2\pi B$ bandwidth covers **90%** of the signal?

Signal Bandwidth – Example 8

Solution: Given that,

$$W = \int_{-\infty}^{\infty} \left| \text{rect} \left(\frac{t}{\tau} \right) \right|^2 dt = \int_{-\frac{\tau}{2}}^{\frac{\tau}{2}} dt = \tau$$

As, $\Omega = \frac{2\pi}{\tau}$ is $r\%$ bandwidth of the signal, where r satisfies,

$$\frac{1}{2\pi} \int_{-\frac{2\pi}{\tau}}^{\frac{2\pi}{\tau}} \tau^2 \text{sinc}^2 \left(\frac{\omega\tau}{2} \right) d\omega = r\tau$$

Signal Bandwidth – Example 8

After change of variable as

$$x \equiv \frac{\omega\tau}{2}$$

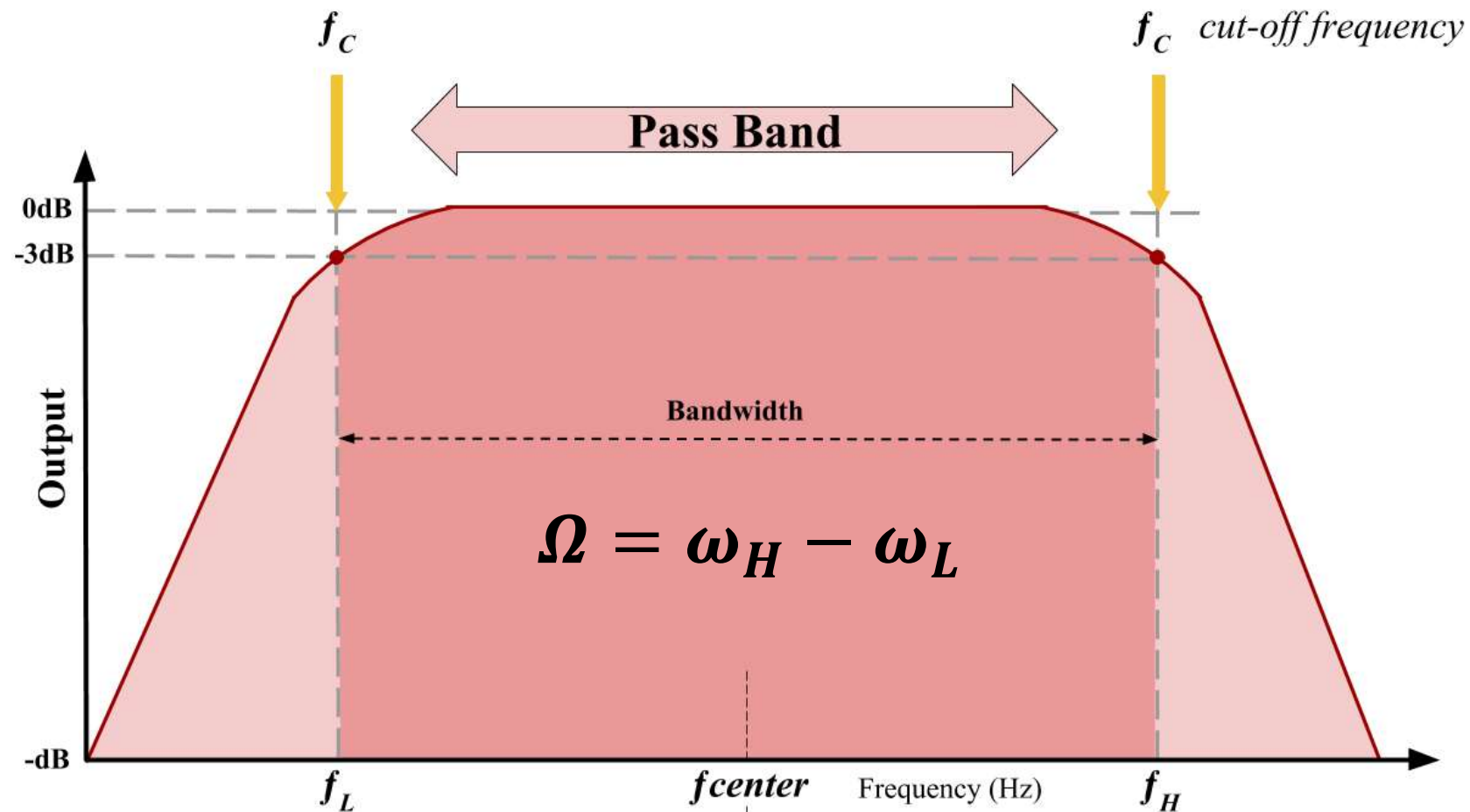
and using even integration,

$$r = \frac{2}{\pi} \int_{x=0}^{\pi} \text{sinc}^2(x) dx \approx 0.903$$

- It can be seen that $\Omega = \frac{2\pi}{\tau}$ corresponding to the frequency of first null in the energy spectrum of the signal $\text{rect}\left(\frac{t}{\tau}\right)$, is the 90.3% bandwidth of the signal

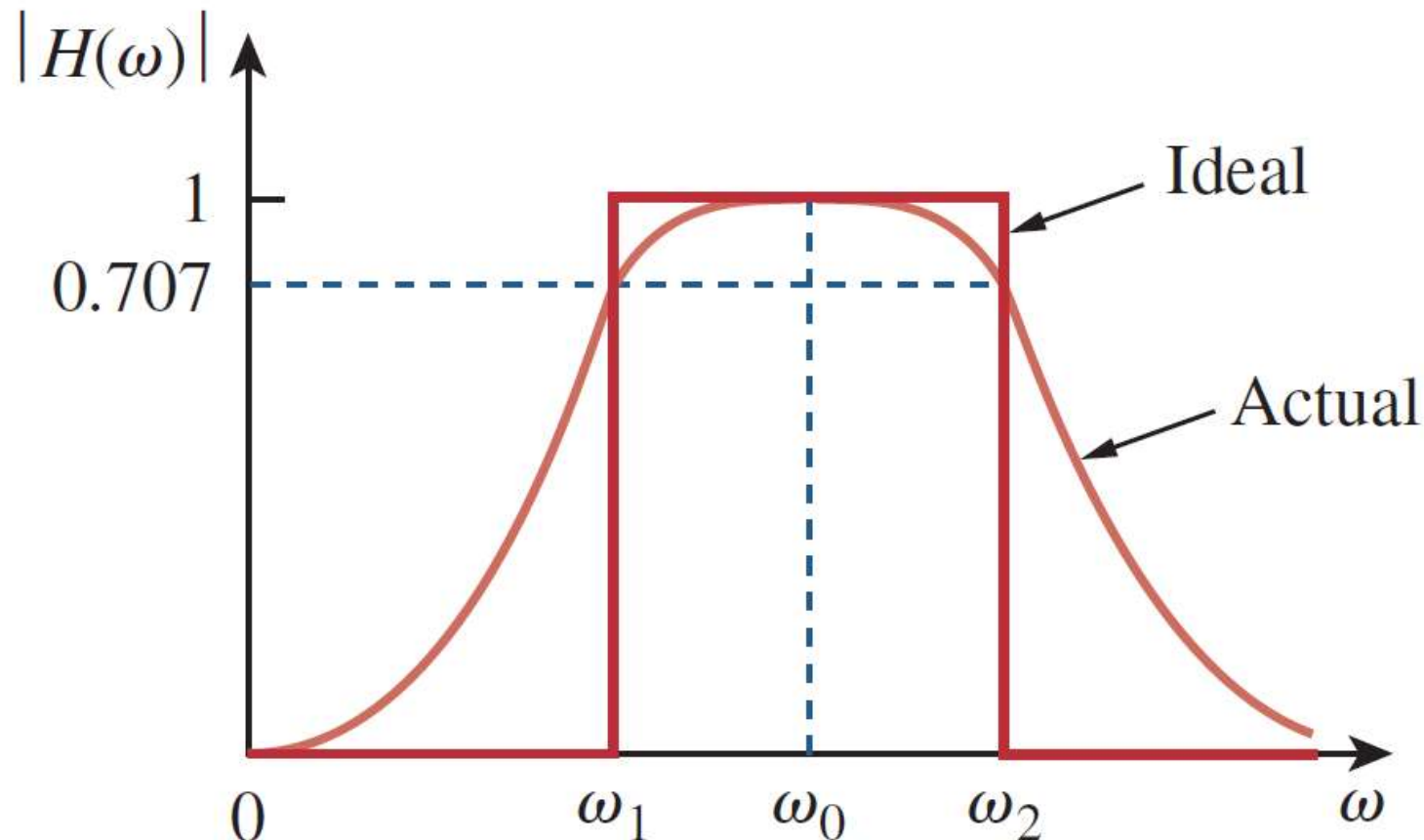
Signal Bandwidth –PB Bandwidth

- The difference between higher frequency spectrum and lower frequency spectrum is passband bandwidth



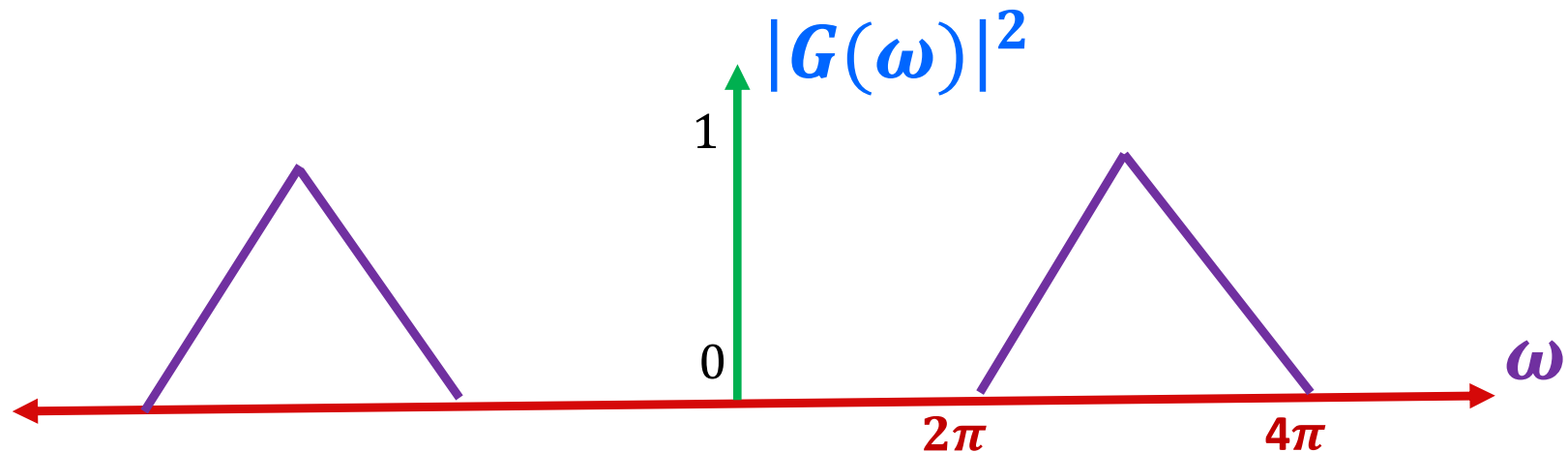
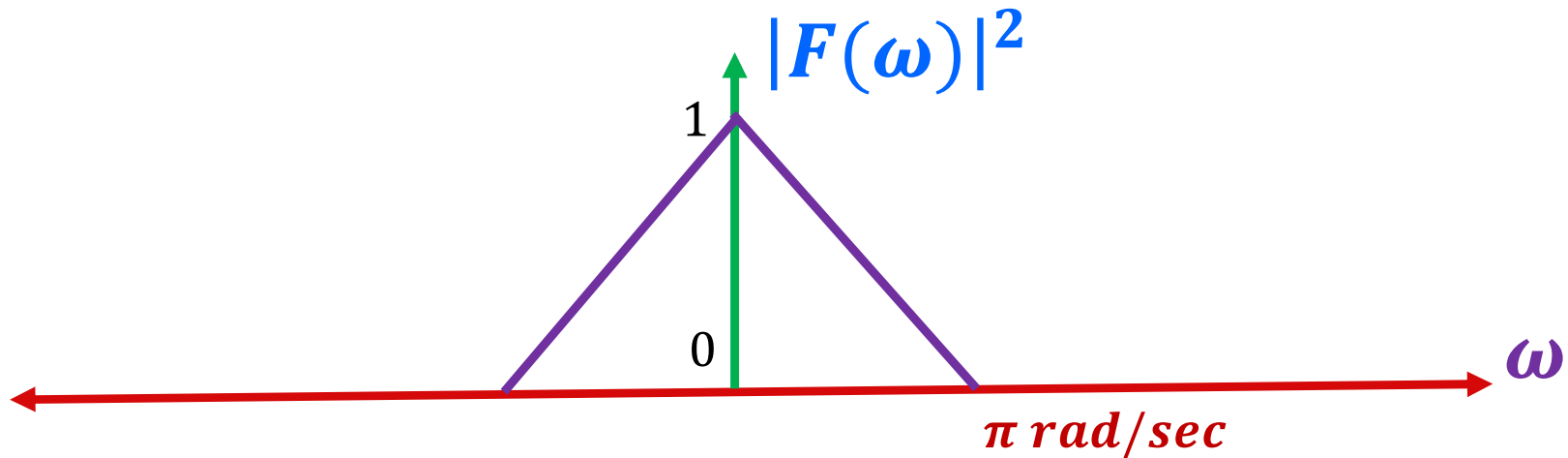
Signal Bandwidth –PB Bandwidth

- A **bandpass filter** is designed to pass all frequencies within a band of frequencies $\omega_1 < \omega < \omega_2$



PB Bandwidth – Example 12

Question: Find the 95% of the bandwidth of $f(t)$ and $g(t)$ for energy spectra shown below?



PB Bandwidth – Example 12

Solution: According to Parseval's theorem,

➤ Energy W of the signal $f(t)$ is simply area under the curve *scaled by* $\frac{1}{2\pi}$

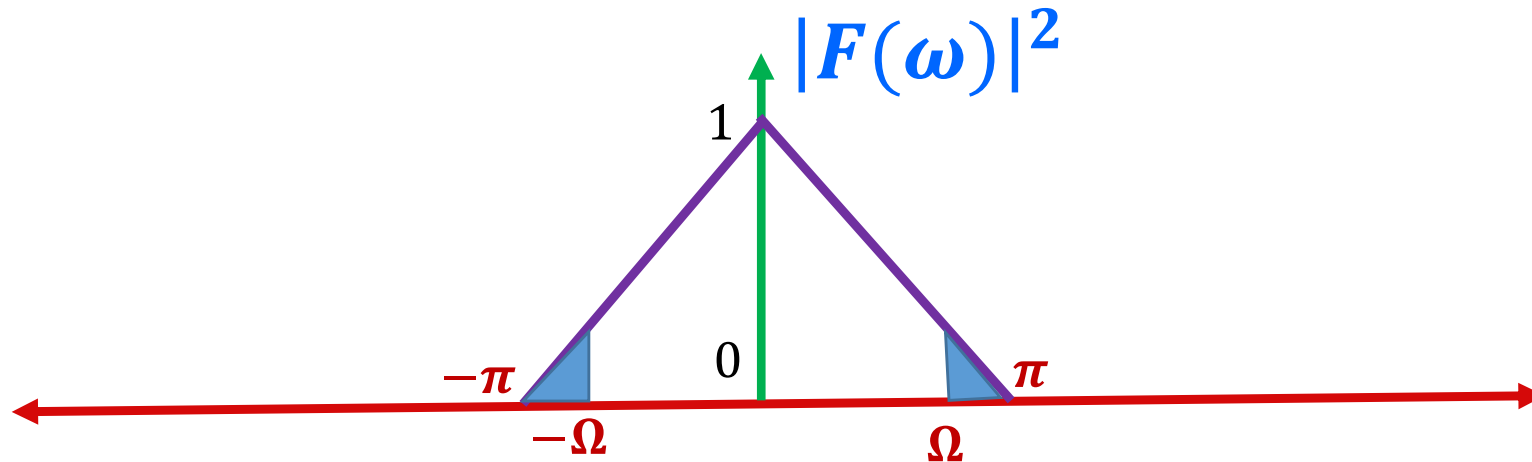
➤ Using the formulae,

$$W = \frac{1}{2\pi} \frac{2\pi \times 1}{2} = \frac{1}{2}$$

➤ To determine 95% bandwidth, we compute the signal energy *outside* signal bandwidth (i.e. $|\omega| > \Omega$)

➤ Set this equal to 5% of W ($0.05W = 0.025$)

PB Bandwidth – Example 12



- This energy outside $|\omega| > \Omega$ equals $\frac{1}{2\pi}$ times the combined areas of right and left tips of $|F(\omega)|^2$

$$\frac{1}{2\pi} (\pi - \Omega) \frac{\pi - \Omega}{\pi}$$

- Setting this quantity equals to 0.025 yields,

PB Bandwidth – Example 12

$$(\pi - \Omega)^2 = 0.05\pi^2$$

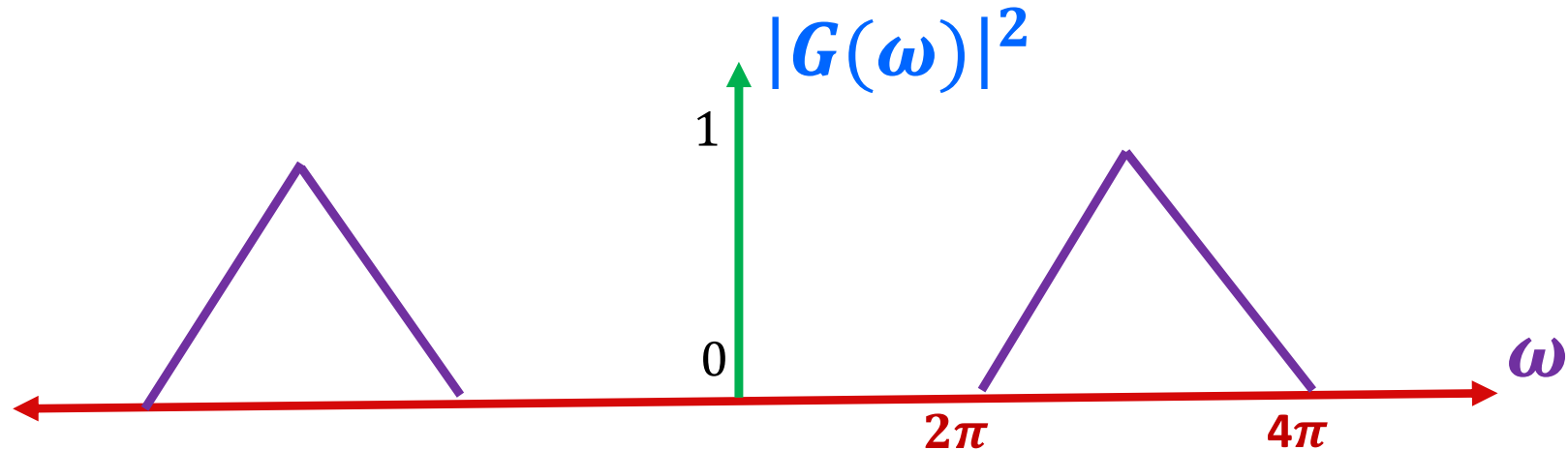
$$\Omega = \pi(1 - \sqrt{0.05}) \approx 0.77\pi \text{ (rad/sec)}$$

- Comparing the two signals energy spectra, it reveals that $|G(\omega)|^2$ is doubles as $|F(\omega)|^2$
- Simply, doubles the bandwidth will work for $|G(\omega)|^2$ 95% bandwidth acquisition

$$\Omega \approx 1.55\pi \text{ (rad/sec)}$$

PB Bandwidth – Example 13

Question: What is 100% bandwidth of $g(t)$?



Solution: It can be seen that,

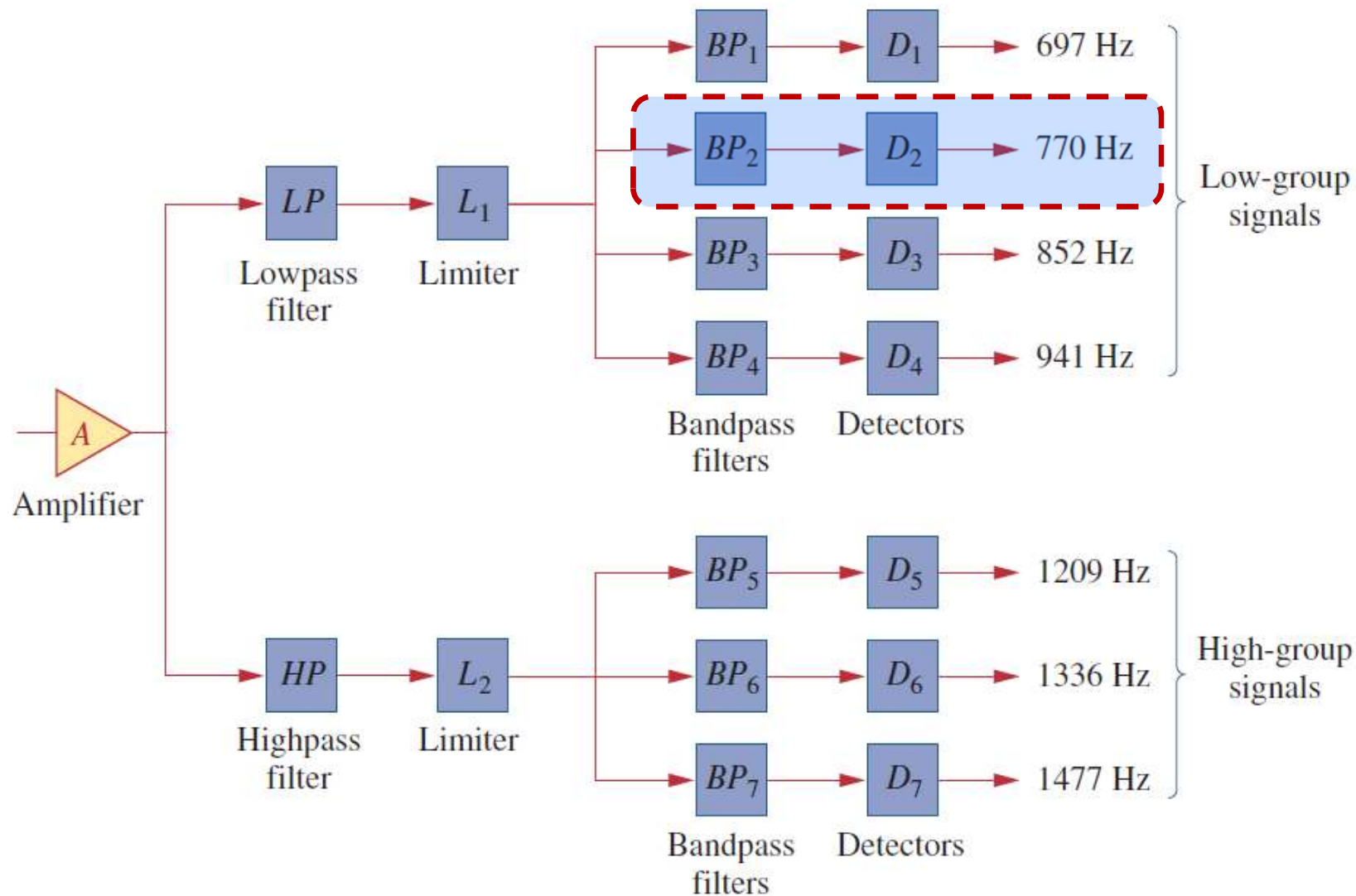
$$\omega_u = 4\pi \text{ \& } \omega_l = 2\pi$$

$$\Omega = \omega_u - \omega_l = 4\pi - 2\pi = 2\pi \text{ rad/sec}$$

or $B = 1 \text{ Hz}$ gives the 100% bandwidth

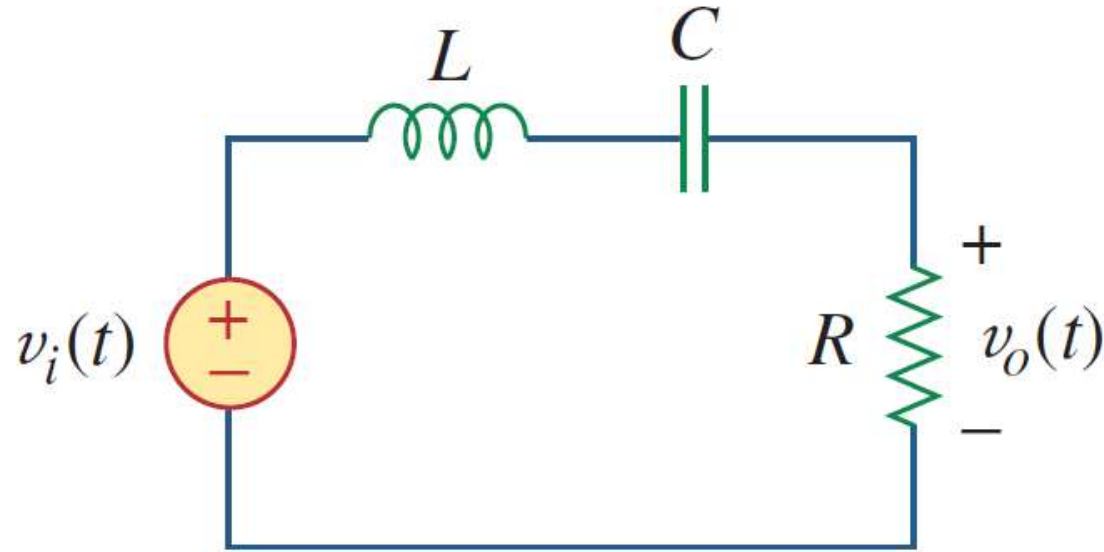
PB Bandwidth – Example 14

Question: Using the standard 600 Ohms resistor used in telephone circuit, design RLC circuit for BP_2 ?



PB Bandwidth – Example 14

Solution: Bandpass filter circuit ($R=600\ \Omega$)



Since BP_2 passes frequencies 697 Hz to 852 Hz and is centered at $f_o = 770\text{ Hz}$, The bandwidth is,

$$B = 2\pi(f_2 - f_1) = 2\pi(852 - 697) = 973.89\text{ rad/s}$$

PB Bandwidth – Example 14

For inductance evaluation, we use relation,

$$L = \frac{R}{B} = \frac{600}{973.89} = 0.616 \text{ H}$$

at resonant frequency, the circuit responds as pure non-resistive so,

$$\begin{aligned} C &= \frac{1}{\omega_0^2 L} = \frac{1}{4\pi^2 f_0^2 L} \\ &= \frac{1}{4\pi^2 \times 770^2 \times 0.616} = 69.36 \text{ nF} \end{aligned}$$

Summary

- The total average power is the sum of the average powers in each harmonically related voltage and current
- Parseval's theorem states that the average power in a periodic signal is the sum of the average power in its dc component and the average powers in its harmonics
- Ratio of second and higher order harmonics to the fundamental harmonic is called total harmonic distortion

Summary

- The *Fourier transform* (FT) is an integral transformation of $f(t)$ from the time domain to the frequency domain
- A sufficient but not necessary condition that $f(t)$ has a Fourier transform is that it be completely integrable
- By using different properties and transform pairs, many non-periodic inputs can be evaluated
- Circuits are solved using FT having basic theorems but initial conditions are not satisfied with FT

Summary

- An aperiodic signal can be expressed in signal energy spectrum instead of power by taking FT over range $[-\infty, \infty]$
- For the case, Parseval's theorem is translated into Rayleigh theorem for infinite energy signals
- The bandwidth refers to the span of upper frequency and lower frequency spectrum
- The energy spectrum falls to one half of the spectral value at DC level

Further reading

1. Ch. 6 (page 212-218), Ch. 7 (page 223-246), E. Kudeki and D. C. Munson, *Analog Signals and Systems*, Prentice Hall, 2008.
2. Ch. 14 (page 650-670), Ch. 17 (page 782-785) and Ch. 18 (page 814-836), C. K. Alexander and M. Sadiku, *Fundamentals of Electric Circuits*, 5th ed., McGraw-Hill, 2013.
3. Ch. 15 (page 755-765, 773-783), J. D. Irwin, and R. M. Nelms. *Basic Engineering Circuit Analysis*, 10th ed. Wiley, 2011.

Preview:

1. Ch. 7 (page 248-258), E. Kudeki and D. C. Munson, *Analog Signals and Systems*, Prentice Hall, 2008.

Homework 9

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

Thank you!