

Answers for "Intro. to Signals & Comm. 2015"

1.a i) $F(\omega) = \int_{-\infty}^{\infty} f(t) e^{-j\omega t} dt$

$$= \int_{-a}^a e^{-j\omega t} dt$$

$$= \frac{e^{-j\omega a} - e^{j\omega a}}{-j\omega}$$

$$= \frac{2 \sin(\omega a)}{\omega}$$

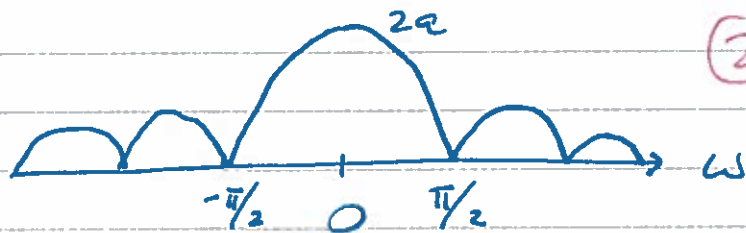
$$= 2a \cdot \text{sinc}(\omega a)$$

total

↓
[2]

(2)

ii) $|F(\omega)|$



(2)

[2]

iii) The system is not realizable⁽¹⁾ in practice because of $f(t) \neq 0$ for $t < 0$. That is, it violates the causality property. The system has a response even before the unit impulse is applied to the system.⁽¹⁾

[2]

1. b. i.

$$\omega_0 = \frac{2\pi}{T_0}$$

(1)

[1]

ii.

iii)

$$\int_0^{T_0} e^{jm\omega_0 t} \cdot e^{-jn\omega_0 t} dt$$

$$= \int_0^{T_0} e^{j(m-n)\omega_0 t} dt$$

(1)

$$= \int_0^{2\pi/\omega_0} e^{j(m-n)\omega_0 t} dt$$

$$= \begin{cases} 0 & m \neq n \\ 2\pi/\omega_0 & m = n \end{cases}$$

(2)

(1)

[1]

[3]

iv.

$$g(t) = \sum_{n=-\infty}^{\infty} D_n e^{jn\omega_0 t}$$

$$\Rightarrow \int_0^{T_0} g(t) e^{-jm\omega_0 t} dt = \int_0^{T_0} \sum_{n=-\infty}^{\infty} D_n e^{jn\omega_0 t} \cdot e^{-jm\omega_0 t} dt$$

(1)

$$\Rightarrow \int_0^{T_0} g(t) e^{-jm\omega_0 t} dt = \sum_{n=-\infty}^{\infty} D_n \cdot \int_0^{T_0} e^{j(n-m)\omega_0 t} dt$$

$$\Rightarrow T_0 D_m = \int_0^{T_0} g(t) e^{-jm\omega_0 t} dt$$

$$\Rightarrow D_m = \frac{1}{T_0} \int_0^{T_0} g(t) e^{-jm\omega_0 t} dt$$

(2)

for $\forall m$

[3]

1. b. v. Each signal component $e^{jn\omega t}$ represents a sinusoidal carrier at frequency $n\omega$. [2]

vi. The Fourier series coefficient D_n represents how closely the signal $g(t)$ resembles the component $e^{jn\omega t}$. [2]

1 c. i. $\phi_{AM}(t) = A \cos \omega_c t + A m(t) \cos \omega_c t$ [2]

ii. The AM essentially translates the signal $m(t)$ at baseband to the frequency band surrounding the carrier frequency ω_c . [2]

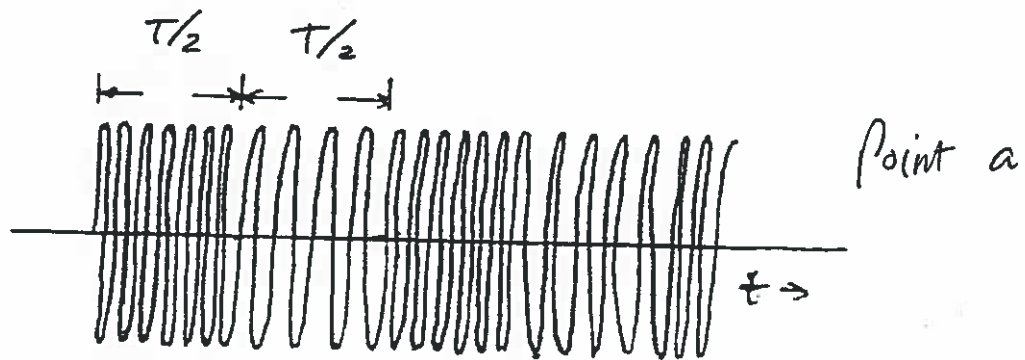
iii. The bandwidth of the AM signal is $2B$ Hz. [2]

iv. $P_c = \frac{A^2}{2}$ [2]

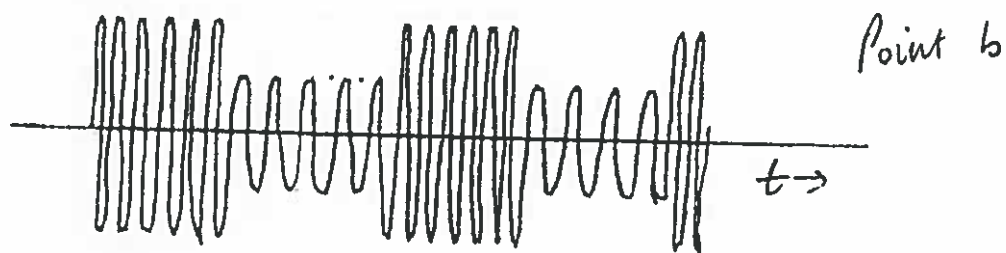
v. $P_s = \frac{E[m^2(t)]}{2} = \lim_{T \rightarrow \infty} \int_{-T/2}^{T/2} m^2(t) \cos^2 \omega_c t \, dt$ [2]

vi. $\eta = \frac{P_s}{P_s + P_c} = \frac{\frac{E[m^2(t)]}{2}}{\frac{E[m^2(t)]}{2} + \frac{A^2}{2}}$

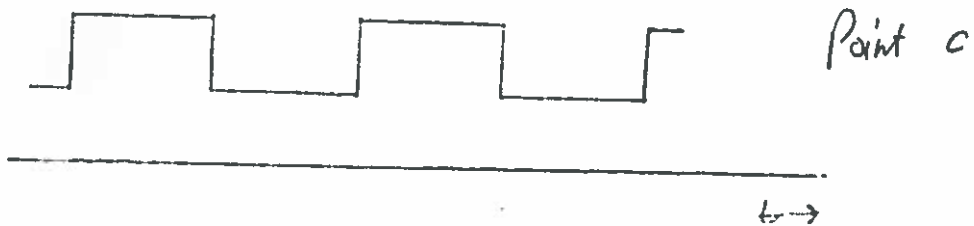
$\eta = \frac{E[m^2(t)]}{E[m^2(t)] + A^2}$ [2]



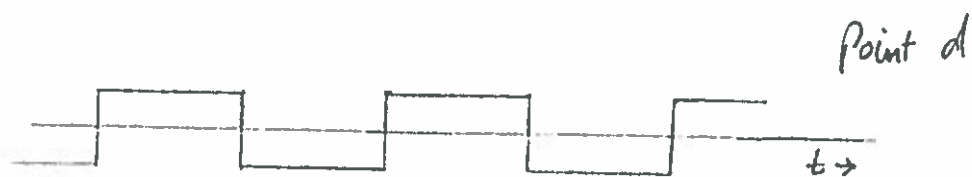
(3)



(3)



(2)



(2)

[10]

2. a. i.

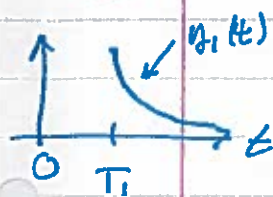
$$y_1(t) = \int_{-\infty}^{\infty} x(u) h(t-u)$$

$$y_1(t) = \int_{-\infty}^{\infty} h(u) x(t-u) du \quad [3]$$

(2)

ii.

$$y_1(t) = \int_{-\infty}^{\infty} \delta(u-T_1) x(t-u) du$$



$$y_1(t) = x(t-T_1)$$

(2)

due to the property of $\delta(\cdot)$

[4]

iv. The physical effects due to $h_1(t)$ and $h_2(t)$ on the input $x(t)$ are to delay the input by T_1 and T_2 time units, respectively.

[2]

iv.

$$h(t) = h_1(t) + h_2(t) \quad [2]$$

v.

$$y(t) = y_1(t) + y_2(t)$$

$$= x(t) * [h_1(t) + h_2(t)] \quad (2)$$

$$\Rightarrow Y(\omega) = X(\omega) \cdot [H_1(\omega) + H_2(\omega)] \quad (2)$$

[4]

2. b. i. $\mathcal{F}[f(t-T)] = \int_{t=-\infty}^{\infty} f(t-T) e^{-j\omega t} dt$ (2)

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

$$\Rightarrow \mathcal{F}[f(t-T)] = e^{-j\omega T} F(\omega) \quad (2) \quad [4]$$

ii. The magnitude of the spectrum for $f(t-T)$ does not change, but the phase of the transform does change according to $e^{-j\omega T}$. (2)

iii. $\mathcal{F}[f(at)] = \int_{t=-\infty}^{\infty} f(at) e^{-j\omega t} dt$ (2)

$$= \frac{1}{a} \int_{t=-\infty}^{\infty} f(at) e^{-j(\frac{\omega}{a})at} d(at)$$

$$= \frac{1}{a} \int_{t'=-\infty}^{\infty} f(t') e^{-j(\frac{\omega}{a})t'} dt'$$

with $t' = at$

$$\Rightarrow \mathcal{F}[f(at)] = \frac{1}{a} F\left(\frac{\omega}{a}\right) \quad (2) \quad [4]$$

2.b. iv. For $a > 1$, $f(at)$ is a "compressed" version of $f(t)$ in time. That is, $f(at)$ changes more rapidly when compared with $f(t)$. ~~But~~ In that case, $F(\frac{\omega}{a})$ is "stretched" in ω .

That is, one can stretch $F(\omega)$ in ω in order to obtain $F(\omega/a)$. As a result, more and more energy of $f(at)$ is located in higher frequency regions. (2)

For $a < 1$, the opposite is true. That is, $f(at)$ can ~~be~~ be obtained by stretching $f(t)$ in time t . Then $F(\omega/a)$ is "squeezed" in ω when compared with $F(\omega)$. That is, $F(\omega/a)$ has more and more energy located at lower frequency regions when compared with $F(\omega)$. (1) [3]

v. When a signal has rapid changes in time, the ~~time~~ transform has higher and higher components in high frequency regions. When the signal changes slowly in time, most of its energy will be associated with low frequencies.

[2]

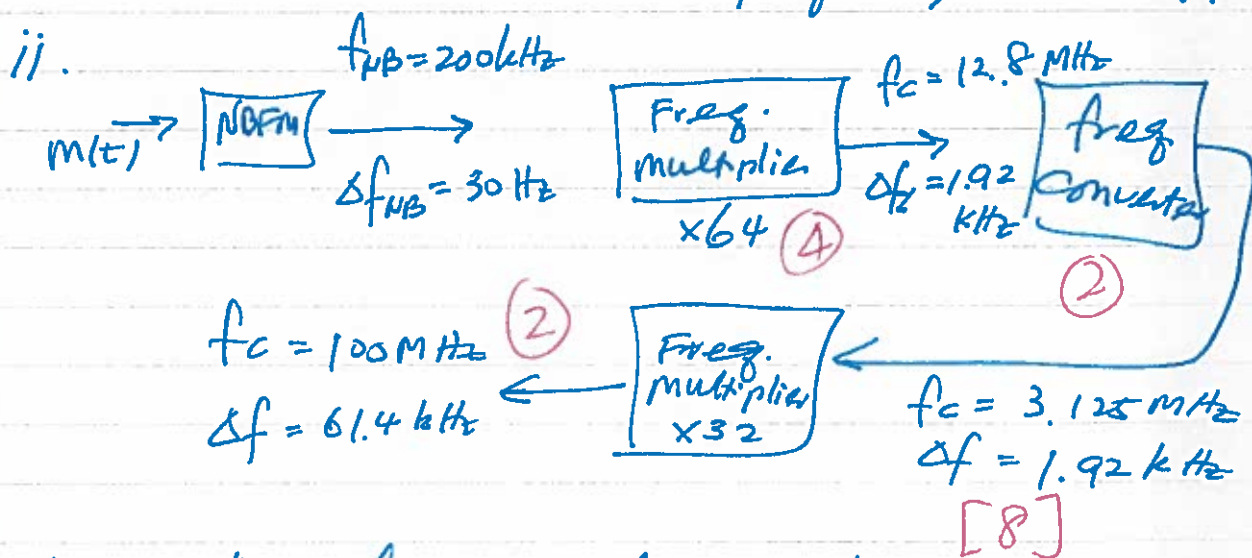
3.a. i. $y(t) = a \phi(t) + b \phi^2(t)$

$$\Rightarrow y(t) = aA \cos[2\pi f_{NB}t + \Delta f_{NB}] + bA^2 \cos^2[2\pi f_{NB}t + \Delta f_{NB}] \quad (2)$$

$$\Rightarrow y(t) = a \cdot A \cos[2\pi f_{NB}t + \Delta f_{NB}] + \frac{bA^2}{2} + \frac{bA^2}{2} \cos[2 \cdot 2\pi f_{NB}t + 2\Delta f_{NB}] \quad (2)$$

[4]

Obtain the carrier freq. and frequency deviation.



iii. Oscillator frequencies for converter

$$9.675 \text{ MHz} = 12.8 \text{ MHz} - 3.125 \text{ MHz} \quad (2)$$

Converter is ^{to} multiply a sinusoidal carrier so that the frequency shift can be effected

$$\begin{aligned} & \cos(\omega_1 t) \cos(\omega_2 t) \\ &= \frac{1}{2} [\cos(\omega_1 - \omega_2)t + \cos(\omega_1 + \omega_2)t] \end{aligned} \quad (4) \quad (6)$$

$$\text{Where } \omega_1 = 12.8 \times 2\pi \text{ rad/sec}$$

$$\omega_1 - \omega_2 = 3.125 \times 2\pi \text{ rad/sec}$$

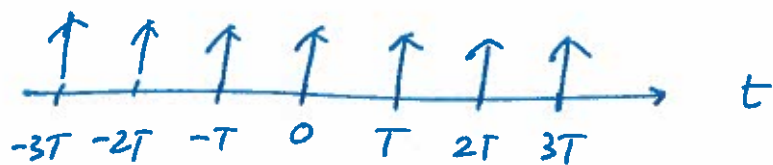
$$\Rightarrow \text{oscillator freq } \omega_2 = 12.8 - 3.125 \\ = 9.675 \text{ MHz}$$

3. b. i. The signaling rate

$$f_s \leq 2B$$

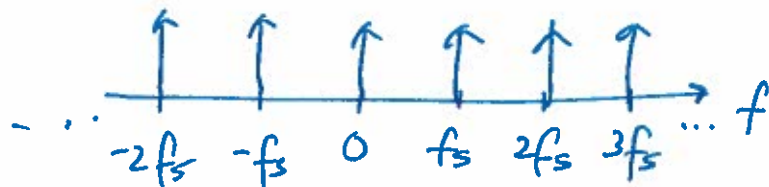
[2]

ii. The samples of signal can be interpreted as applying the sampling of pulse train onto a given signal:



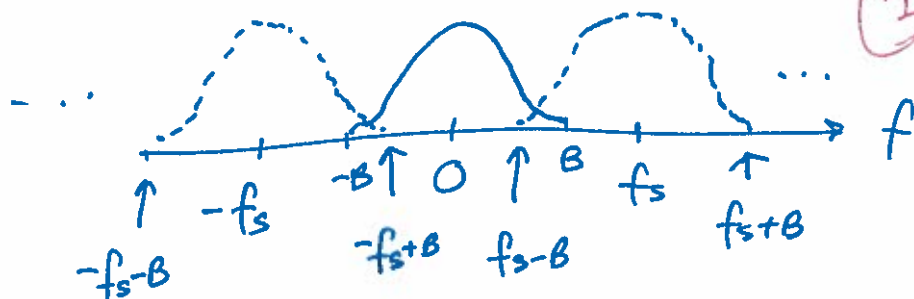
(2)

which has a spectrum of Fourier transform



(2)

As a result, The signal samples ^{of the signal} represent a spectrum of



(3)

[7]

From the above diagram,

if $B \geq f_s - B$,

the replica of the baseband spectrum of the original signal do not overlap. That means the original signal has a bandwidth of B Hz or less, which can be supported for transmission by a channel of B Hz. Otherwise, the original signal has components beyond B Hz, which cannot all pass through the channel of B Hz — ~~correct~~
Correct reception is impossible.

Overall, the signal rate $f_s \leq 2B$. ✓

iii.

$$C = 2B \log_2 M \quad [3]$$