

- 1.1 (a) One dimensional, multichannel, discrete time and digital.
 This signal is generated by multiple sources at certain specific time. Thus it is discrete-time and multichannel signal. Since this discrete-time signal has a set of discrete values, it is digital. Because it is also a function of a single independent variable, it is one-dimensional signal.
- (b) Multi-dimensional, single channel, continuous time and analog.
 (c) One-dimensional, single channel, continuous time and analog.
 (d) One-dimensional, single channel, continuous time and analog.
 (e) One-dimensional, ~~single~~ multichannel, discrete time and digital.

1.7

(a) $F_{max} = 10 \text{ kHz} \Rightarrow F_s > 2F_{max} = 20 \text{ kHz}$. This is based on Sampling Theorem.

(b) Since $F_s = 8 \text{ kHz} < 20 \text{ kHz}$, there is aliasing.

Therefore, the frequency content above the sampling rate F_s gets folded about $F_{fold} = \frac{1}{2} F_s = 4 \text{ kHz}$. So the frequency $F_1 = 5 \text{ kHz}$ gets folded to $F_a = 3 \text{ kHz}$.

Note: We can also calculate the aliasing frequency F_a by using the following equation: $F_a = |k \cdot F_s - F_1|$, where k is the integer to get minimum F_a . e.g. in this question, $F_a = |8 \cdot k - 5|$, by choosing $k=1$, we can get $F_a = 3 \text{ kHz}$.

(c) Following the same method illustrated in (b), we can get the frequency $F_2 = 9 \text{ kHz}$ get folded to $F_a = 1 \text{ kHz}$.

$$1.8) F_s = 100 \text{ Hz}$$

$$\text{a) Nyquist frequency} = 2F_{\max}$$

$$\Rightarrow \text{Nyquist frequency} = 200 \text{ Hz}$$

$$\text{b) } F_s = 250 \text{ samples/s}$$

The maximum frequency that can be recovered is $\frac{250}{2} = 125 \text{ Hz}$

$$1.9) x_a(t) = \sin(480\pi t) + 3 \sin(720\pi t), F_s = 600 \text{ Hz}$$

a) The maximum frequency in this signal is radians is

$$\omega_{\max} = 720\pi \text{ radians/sec}$$

$$\text{since } \omega_{\max} = 2\pi F_{\max} \Rightarrow F_{\max} = 360 \text{ Hz}$$

$$\therefore \text{The Nyquist rate} = 2F_{\max} = 720 \text{ Hz}$$

$$\text{b) The folding frequency } F_{\text{fold}} = F_s/2 \Rightarrow F_{\text{fold}} = \frac{600}{2} = 300 \text{ Hz}$$

c) The sampling of a continuous signal to a discrete signal is described by:

$$x(n) = x_a(nT), \text{ we know } T = \frac{1}{F_s} = 600 \text{ Hz}$$

$$\Rightarrow x(n) = \sin\left(\frac{480\pi}{600}n\right) + 3\sin\left(\frac{720\pi}{600}n\right)$$

$$= \sin\left(\frac{4\pi}{5}n\right) - 3\sin\left(\frac{4\pi}{5}n\right)$$

$$\Rightarrow x(n) = -2\sin\left(\frac{4\pi}{5}n\right)$$

so $\omega = 2\pi f_n \Rightarrow \omega = \frac{4\pi}{5}$ radians/sample

d) We know the sampling frequency is 600 times/second

since $f = \frac{F}{F_s}$

$$\Rightarrow y_d(t) = -2\sin\left(\frac{4\pi}{5} \times 600t\right) = -2\sin(480\pi t)$$

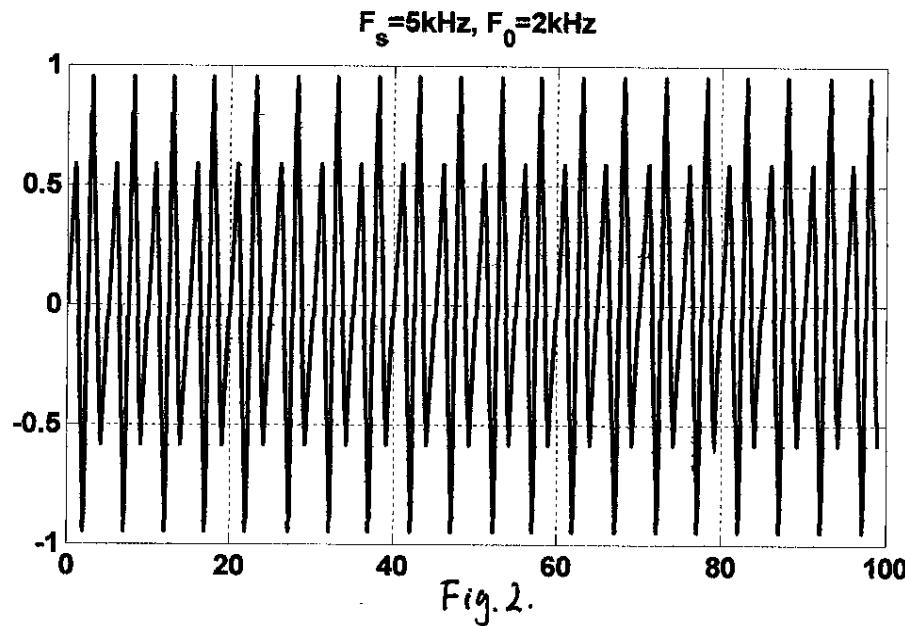
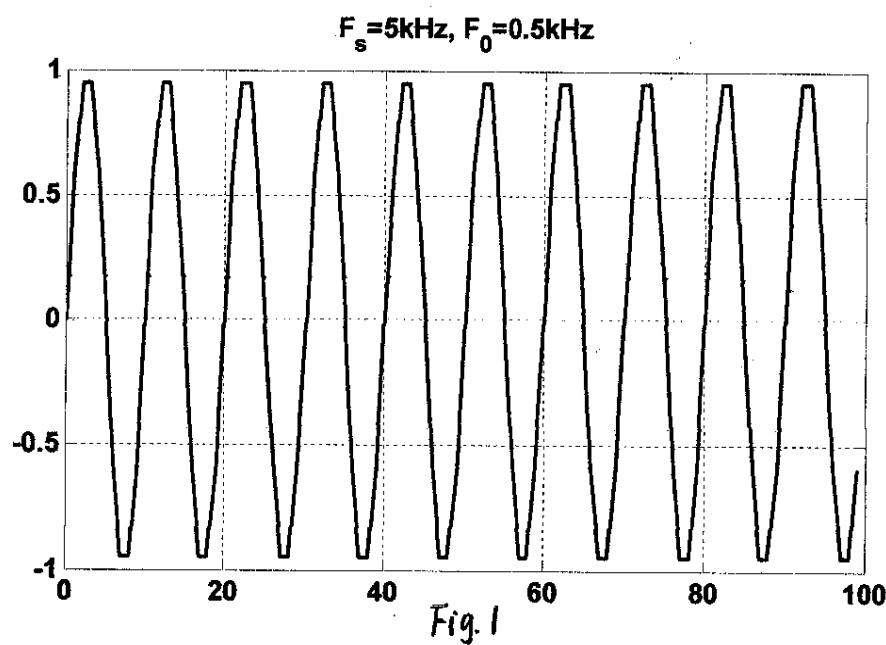
1.15

P₂

(a) The plot in Fig. 1 is the mirror-image of the plot in Fig.4.
 The plot in Fig.2 is the mirror-image of the plot in Fig.3.

When $F_0 = 3\text{ kHz}$ and 4.5 kHz , $F_s < 2F_0$, therefore this is aliasing.

$F_{\text{fold}} = \frac{F_s}{2} = 2.5\text{ kHz}$, thus $F_0 = 3\text{ kHz}$ is aliased to 0.5 kHz and $F_0 = 4.5\text{ kHz}$ is aliased to 2 kHz .



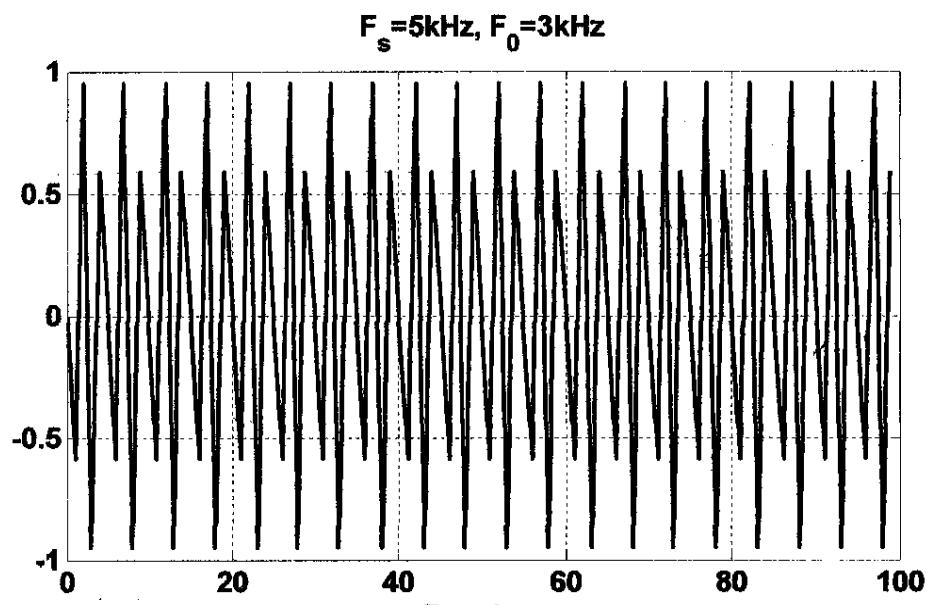
P₃

Fig.3

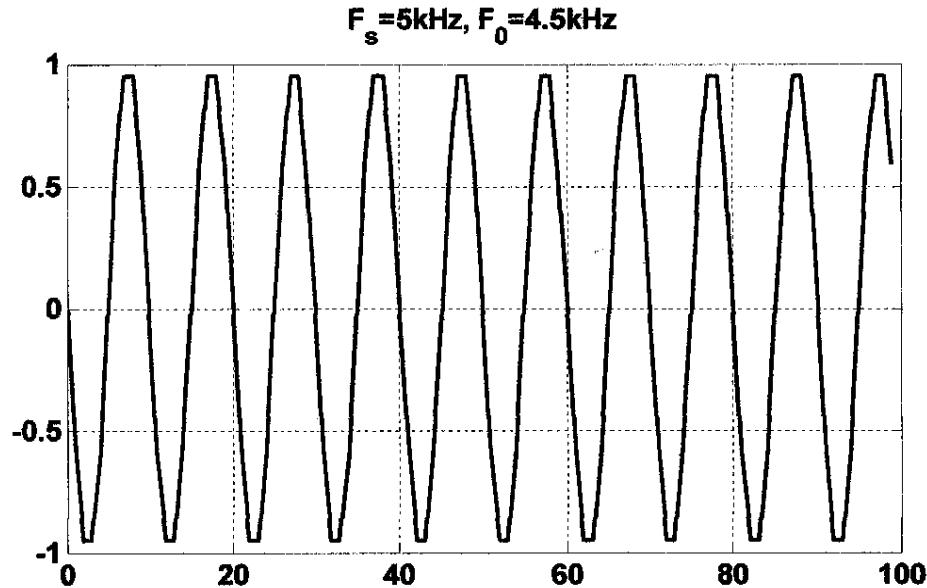


Fig.4

1.15

$$(b) x(n) = \sin(2\pi \frac{F_0}{F_s} n) = \sin(2\pi \frac{2}{50} n) = \sin(\frac{2\pi}{25} n)$$

Thus $f_0 = \frac{1}{25}$ and the plot is shown in Fig. 5.

2. By taking the even numbered samples, the sampling frequency is reduced to half of F_s , i.e. 25Hz, which is still larger than the Nyquist rate $2F_0$. Thus, there is no aliasing, and $y(n)$ is still a sinusoidal signal. $y(n) = \sin(2\pi \frac{F_0}{F_s} n) = \sin(2\pi \frac{2}{25} n) = \sin(\frac{4\pi}{25} n)$ is plotted in Fig. 6, and its frequency is $f_0 = \frac{2}{25}$.

$$F_0 = 2\text{kHz}, F_s = 50\text{kHz}$$

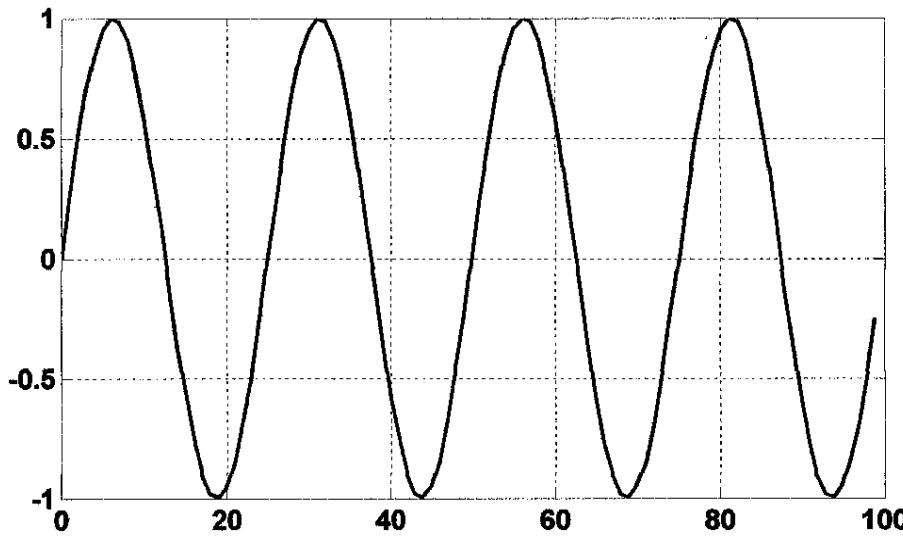


Fig. 5

$$F_0 = 2\text{kHz}, F_s = 25\text{kHz}$$

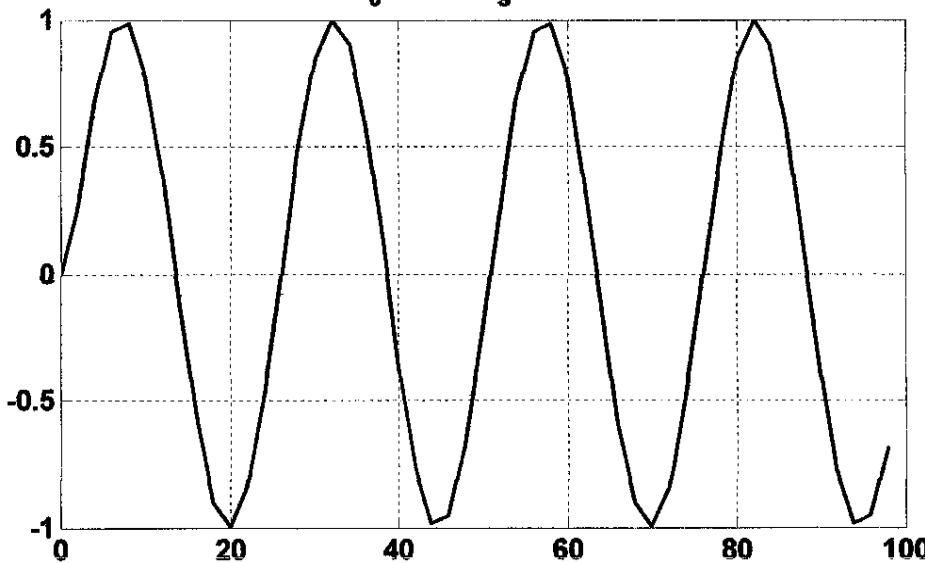


Fig. 6