EE2023 TUTORIAL 4 (SOLUTIONS)

Solution to Q.1

$$\widetilde{x}(t) = \underbrace{\text{rect}\left(\frac{t - 0.475}{0.45}\right)}_{x(t)} \cdot \underbrace{\sum_{n = -\infty}^{\infty} \delta(t - 0.2n)}_{\Xi_{0.2}(t)}$$
$$= \delta(t - 0.4) + \delta(t - 0.6)$$

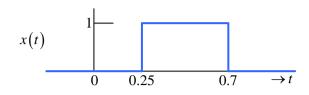
$$\Im\{\delta(t-\varsigma)\} = \exp(-j2\pi f\varsigma)$$

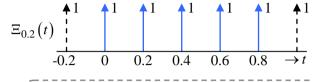
$$\tilde{X}(f) = \exp(-j2\pi f 0.4) + \exp(-j2\pi f 0.6)$$

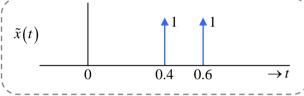
$$= \begin{cases}
\cos(0.8\pi f) - j\sin(0.8\pi f) + \\
\cos(1.2\pi f) - j\sin(1.2\pi f)
\end{cases}$$

$$= \begin{cases}
\left[\cos(0.8\pi f) + \cos(1.2\pi f)\right] - \\
j\left[\sin(0.8\pi f) + \sin(1.2\pi f)\right]
\end{cases}$$

$$= 2\cos(\pi f)\cos(0.2\pi f) - j2\sin(\pi f)\cos(0.2\pi f)$$

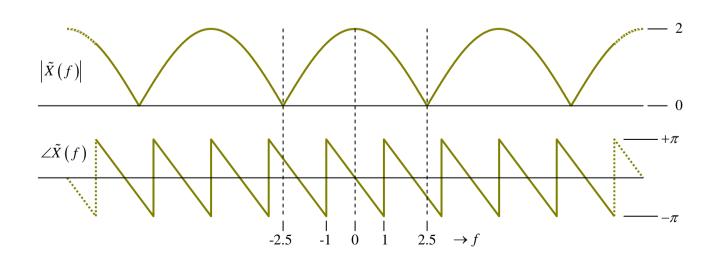


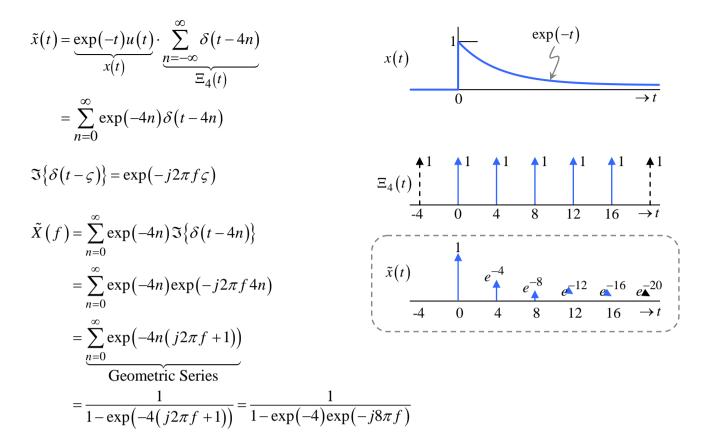




$$\left| \tilde{X}(f) \right|^2 = 4\cos^2(\pi f)\cos^2(0.2\pi f) + 4\sin^2(\pi f)\cos^2(0.2\pi f)$$
$$= 4\cos^2(0.2\pi f)$$

- $\bullet \quad \left| \tilde{X}(f) \right| = 2 \left| \cos(0.2\pi f) \right|$





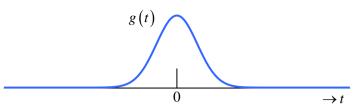
When a signal is sampled in the time domain, the spectrum of the sampled signal is periodic with a period equal to the sampling frequency, which in this case is 0.25 Hz.

Thus, to show that $\tilde{X}(f)$ is periodic, all we need to show is $\tilde{X}(f+0.25) = \tilde{X}(f)$ as follows:

$$\tilde{X}(f+0.25) = \frac{1}{1 - \exp(-4)\exp(-j8\pi(f+0.25))}$$

$$= \frac{1}{1 - \exp(-4)\exp(-j8\pi f)}\underbrace{\exp(-j2\pi)}_{1}$$

$$= \frac{1}{1 - \exp(-4)\exp(-j8\pi f)} = \tilde{X}(f)$$



Rewrite $x(t) = \sum_{n=-\infty}^{\infty} g(t - nT_p)$ in convolution form:

$$x(t) = g(t) * \sum_{n=-\infty}^{\infty} \delta(t - nT_p) \qquad (*)$$

Applying the 'Convolution' property of the Fourier transform to (*):

$$X(f) = G(f) \cdot \frac{1}{T_p} \sum_{k=-\infty}^{\infty} \delta\left(f - \frac{k}{T_p}\right) \cdots (**)$$

Conclusion:

X(f) can be obtained by sampling $G(f)/T_p$ in the frequency-domain at regular spacings of $1/T_p$ Hz.

Relationship between c_k and G(f)

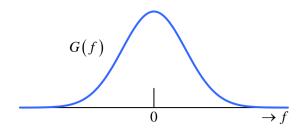
Rewrite (**) as:
$$X(f) = \sum_{k=-\infty}^{\infty} \frac{1}{T_p} G\left(\frac{k}{T_p}\right) \delta\left(f - \frac{k}{T_p}\right)$$

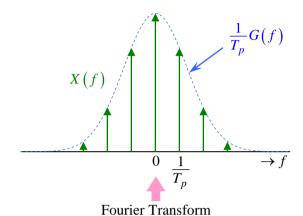
In terms of
$$c_k$$
: $X(f) = \sum_{k=-\infty}^{\infty} c_k \delta\left(f - \frac{k}{T_p}\right)$

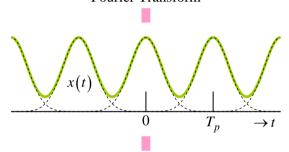
Hence,
$$c_k = \frac{1}{T_p} G(f) \Big|_{f=k/T_p} = \frac{1}{T_p} G\left(\frac{k}{T_p}\right)$$

Uniqueness of a generating function

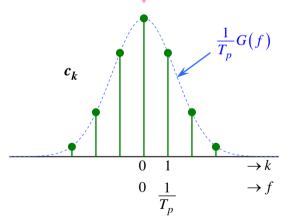
The generating function of a periodic signal is NOT unique. For instance, one period of a periodic signal can be used as its generating function.









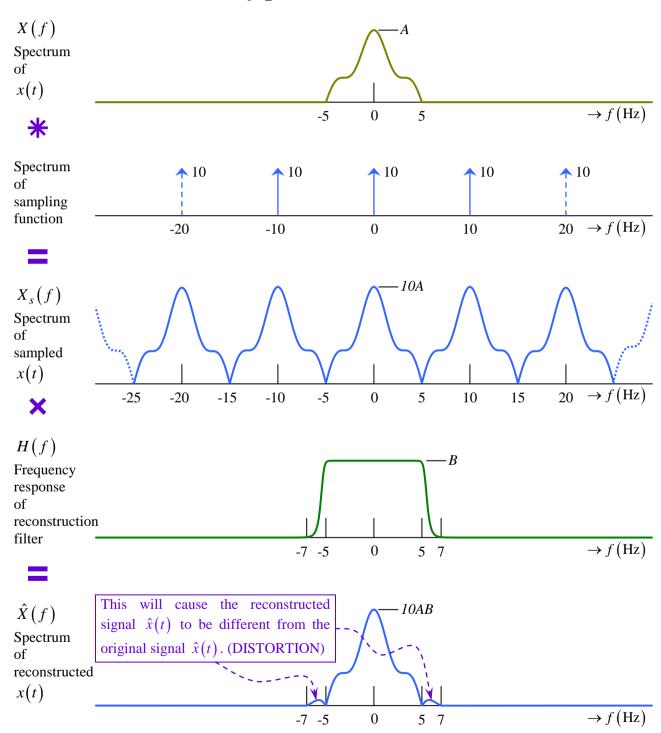


Remarks: From the sampling theorem, we observed that when a signal is sampled at regular time-intervals of $T_s(\sec)$, the spectrum corresponding to the sampled signal is periodic with a period of $f_s = 1/T_s(\operatorname{Hz})$. In this exercise, we showed that when a spectrum is sampled at regular frequency-intervals of $f_p(\operatorname{Hz})$, the signal corresponding to the sampled spectrum is periodic with a period of $T_p = 1/f_p(\sec)$. Hence, sampling a signal leads to a periodic spectrum and sampling a spectrum leads to periodic signal. This phenomenon is due to the convolution property of the Fourier Transform and the fact that $\left[\Xi_T(t) \rightleftarrows \frac{1}{T}\Xi_{1/T}(f)\right]$.

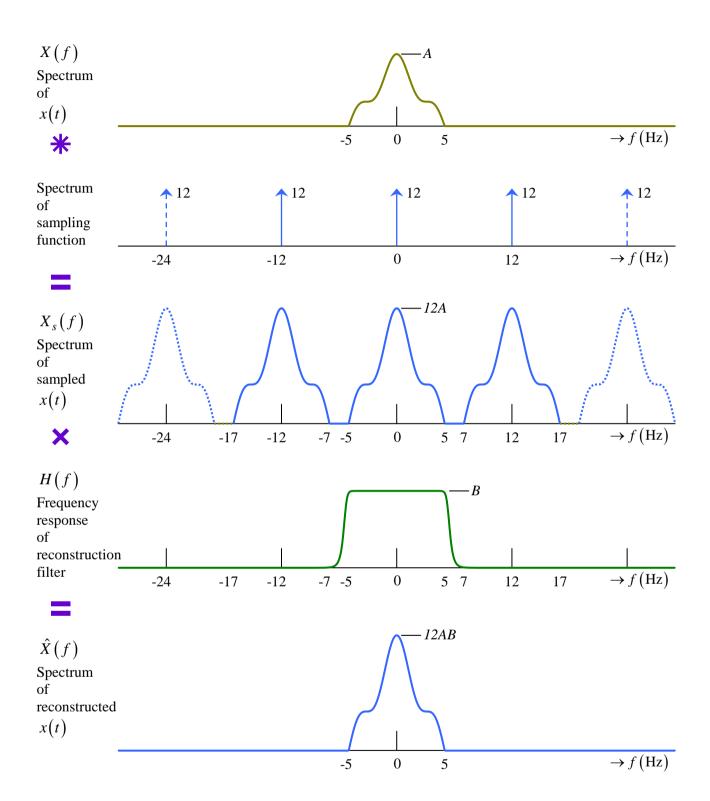
- Nyquist sampling frequency: $2 \times 5 = 10 \text{ Hz}$
- Recommended sampling frequency: 12 Hz

The excess 2 Hz is needed to prevent adjacent spectral images from contributing to the reconstruction process. See illustration below.

Sampling frequency: 10 Hz (Nyquist Rate)

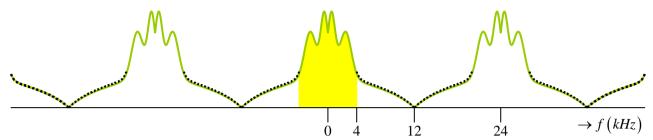


Sampling frequency: 12 Hz



Remarks: Sampling frequencies higher that 12 Hz may be used to achieve the same result. However, high sampling frequency is usually matched by more costly data acquisition, storage and processing requirement.

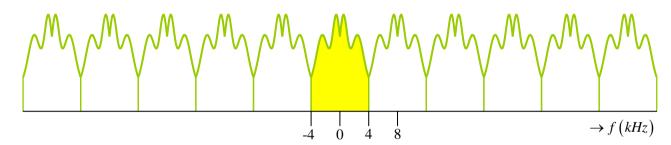
Situation A: $\binom{No \ anti-aliasing \ lowpass \ filter}{and \ no \ frequency \ aliasing}$: $\begin{cases} Sampling \ frequency = 2 \times 12 = 24 \ kHz \end{cases}$



Advantage: No anti-aliasing LPF needed.

Disadvantage: Sampling frequency is higher than necessary to preserve the TQ-Band.

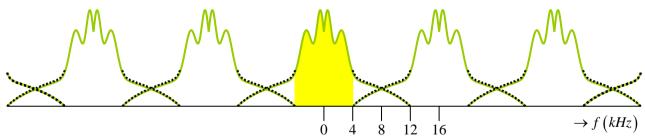
Situation B: Lowest sampling frequency: $\begin{cases} \text{Ideal anti-aliasing filter of bandwidth } 4 \text{ } kHz \\ \text{Sampling frequency} = 2 \times 4 = 8 \text{ } kHz \end{cases}$



Advantage: Lowest possible sampling frequency to preserve the TQ-Band.

Disadvantage: Require anti-aliasing LPF with sharp cutoff.

Situation C: No anti-aliasing lowpass filter: $\left\{\text{Sampling frequency} = 2 \times \left(\frac{12+4}{2}\right) = 16 \text{ kHz}\right\}$



Advantage: Lowest possible sampling frequency to preserve the *TQ-Band* without requiring anti-aliasing LPF with sharp cutoff.

Disadvantage: Sampling frequency is still higher than the minimum needed to preserve the TQ-Band.

This problem illustrates the trade-off between oversampling and the requirement of expensive antialiasing LPF with sharp cutoff frequency.

$$f_S = 2 \times 20 = 40 \text{ kHz}$$

$$\begin{bmatrix} x_{\mathcal{S}}(t) = x(t) \cdot \sum_{n} \mathcal{S}\left(t - \frac{n}{40000}\right) \end{bmatrix} \iff \begin{bmatrix} X_{\mathcal{S}}(f) = X(f) + 40000 \sum_{k} \mathcal{S}(f - 40000k) \\ = 40000 \sum_{k} X(f - 40000k) \end{bmatrix}$$

Specifying $H(f) = \frac{1}{40000} \cdot \text{rect}\left(\frac{f}{40000}\right)$ will lead to $X_s(f)H(f) = X(f)$.

Solution to S.2

(a)
$$\begin{cases} x_s(t) = \sum_{n=-5}^{5} x(5n)\delta(t-5n) \\ & \text{sampling period} \\ \therefore \text{ Sampling frequency: } f_s = \frac{1}{5} = 0.2 \text{ Hz} \end{cases}$$

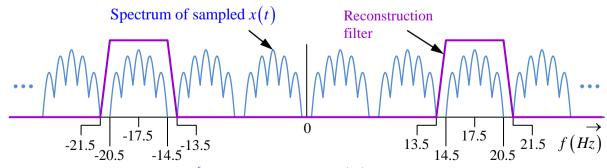
(b) Perfect reconstruction is of x(t) from $x_s(t)$ only possible if x(t) is bandlimited to $f_s/2 = 0.1$, i.e. X(f) = 0; f > 0.1.

Since $X(f) = \Im^{-1}\{\operatorname{tri}(t)\} = \operatorname{sinc}^2(f)$ has infinite frequency extent, perfect reconstruction is not possible due to frequency aliasing.

Solution to S.3

- (a) Nyquist Sampling Frequency: $f_s = 2 \times 20.5 = 41 \, Hz$
- (b) Center frequency: $f_c = 17.5 \ Hz$ Bandwidth: $B = 6 \ Hz$ Possible sampling frequencies : $f_s = 2 f_c / k$; $k = 1, 2, \dots, \lfloor 2 f_c / B \rfloor$

Lowest sampling frequency : $f_s = \frac{2f_c}{|2f_c/B|} = \frac{35}{5} = 7 Hz$



Reconstruction Filter Specs: $\begin{cases} \text{Ideal passband: } 14.5 < |f| < 20.5 \\ \text{Ideal stopband: } |f| < 13.5 \quad or \quad |f| > 21.5 \end{cases}$