

# **EE 123 Discussion Section 6**

## **Sampling**

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Based on slides by Frank Ong

# Announcements

- Submit your Lab 3 results in groups
- Please come to the lab sections next week to pick up your Radios.
- Questions?

# Review of sampling

## Continuous time signal

$$x_c(t) \longleftrightarrow X_c(j\Omega) = \int x_c(t)e^{-j\Omega t} dt$$

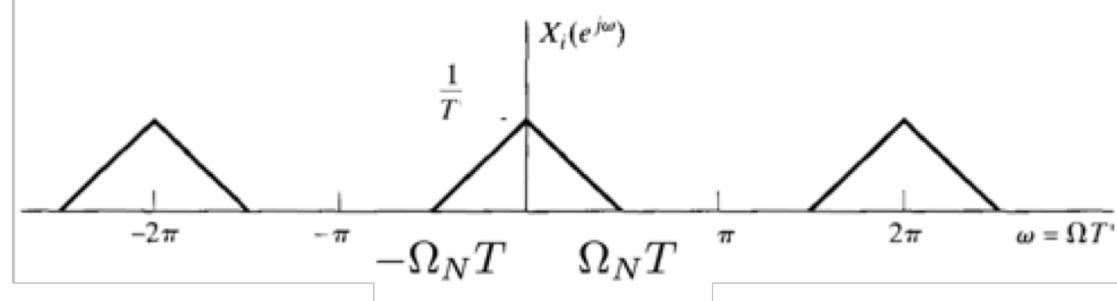
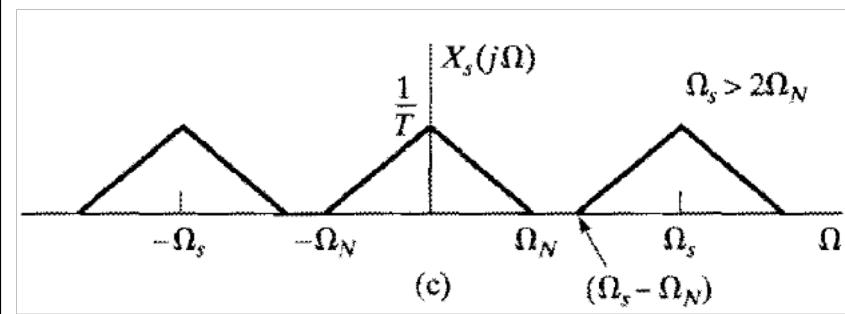
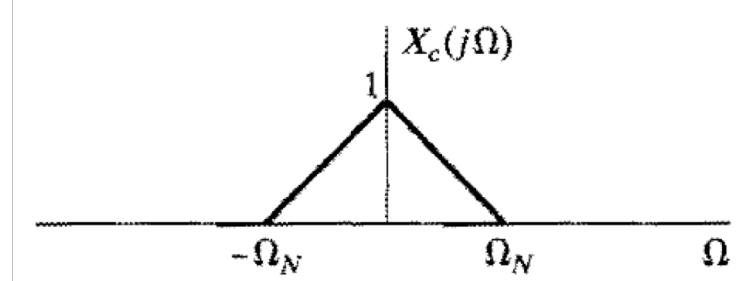
## Continuous time sampling

$$x_s(t) = x_c(t) \sum_{k=-\infty}^{\infty} \delta(t - kT)$$

$$\begin{aligned} X_s(j\Omega) &= \frac{1}{T} \sum_{k=-\infty}^{\infty} X_c(j(\Omega - k\Omega_s)), \quad \Omega_s = \frac{2\pi}{T} \\ &= \sum_{k=-\infty}^{\infty} x_c(nT)e^{-j\Omega T n} \end{aligned}$$

## Discrete time spectrum

$$X(e^{j\omega}) = X_s \left( j \left( \frac{\omega}{T} \right) \right)$$



# Previous HW question 1

Each of the following continuous-time signals is used as the input  $x_c(t)$  for an ideal C/D converter as shown in Figure 4.1 with the sampling period  $T$  specified. In each case, find the resulting discrete-time signal  $x[n]$ .

- (a)  $x_c(t) = \cos(2\pi(1000)t)$ ,  $T = (1/3000)$  sec
- (b)  $x_c(t) = \sin(2\pi(1000)t)$ ,  $T = (1/1500)$  sec
- (c)  $x_c(t) = \sin(2\pi(1000)t) / (\pi t)$ ,  $T = (1/5000)$  sec

## Previous HW question 2a

A continuous-time finite-duration signal  $x_c(t)$  is sampled at a rate of 20,000 samples/s, yielding a 1000-point finite-length sequence  $x[n]$  that is nonzero in the interval  $0 \leq n \leq 999$ . Assume for this problem that the continuous-time signal is also bandlimited such that  $X_c(j\Omega) = 0$  for  $|\Omega| \geq 2\pi(10,000)$ ; i.e., assume that negligible aliasing distortion occurs in sampling. Assume also that a device or program is available for computing 1000-point DFTs and inverse DFTs.

- (a) If  $X[k]$  denotes the 1000-point DFT of the sequence  $x[n]$ , how is  $X[k]$  related to  $X_c(j\Omega)$ ? What is the effective continuous-time frequency spacing between DFT samples?

## Previous HW question 2b

The following procedure is proposed for obtaining an expanded view of the Fourier transform  $X_c(j\Omega)$  in the interval  $|\Omega| \leq 2\pi(5000)$ , starting with the 1000-point DFT  $X[k]$ .

**Step 1.** Form the new 1000-point DFT

$$W[k] = \begin{cases} X[k], & 0 \leq k \leq 250, \\ 0, & 251 \leq k \leq 749, \\ X[k], & 750 \leq k \leq 999. \end{cases}$$

**Step 2.** Compute the inverse 1000-point DFT of  $W[k]$ , obtaining  $w[n]$  for  $n = 0, 1, \dots, 999$ .

**Step 3.** Decimate the sequence  $w[n]$  by a factor of 2 and augment the result with 500 consecutive zero samples, obtaining the sequence

$$y[n] = \begin{cases} w[2n], & 0 \leq n \leq 499, \\ 0, & 500 \leq n \leq 999. \end{cases}$$

**Step 4.** Compute the 1000-point DFT of  $y[n]$ , obtaining  $Y[k]$ .

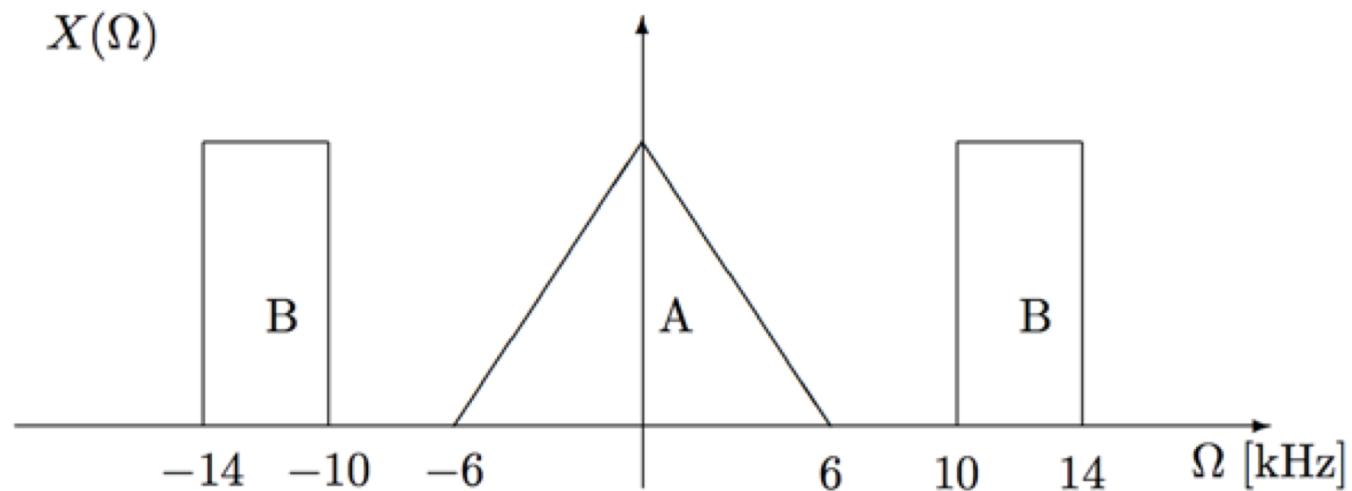
**(b)** The designer of this procedure asserts that

$$Y[k] = \alpha X_c(j2\pi \cdot 10 \cdot k), \quad k = 0, 1, \dots, 500,$$

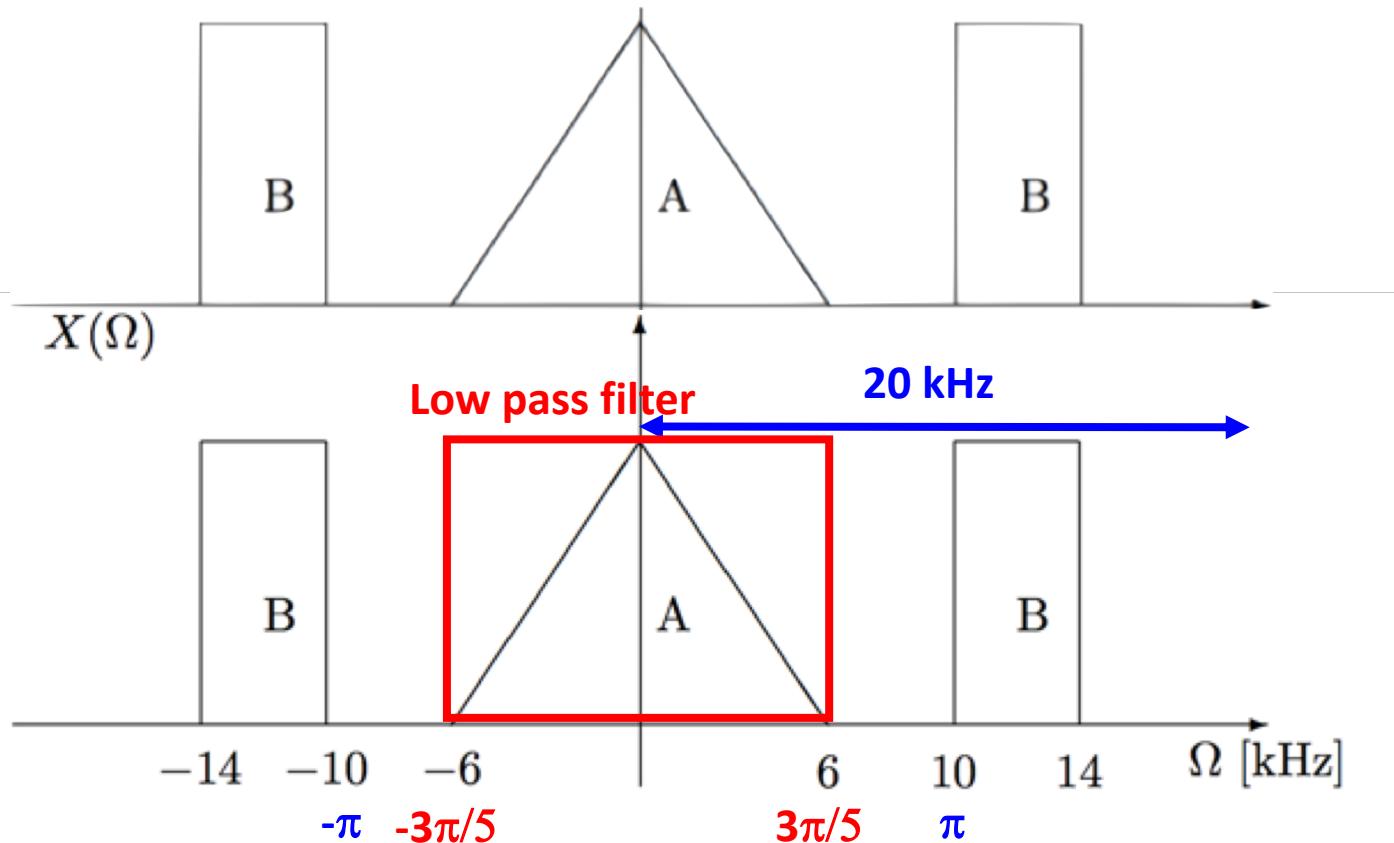
where  $\alpha$  is a constant of proportionality. Is this assertion correct? If not, explain why not.

# Sampling question 1

4. An analog signal, whose spectrum is shown below, is to be processed with a digital filter using ideal C/D and D/C converters (with no analog anti-aliasing filters).
- a) (8 points) What is the minimum sufficient sampling rate to extract portion A of the signal? Sketch the magnitude of the digital filter that would be used at this sampling rate.

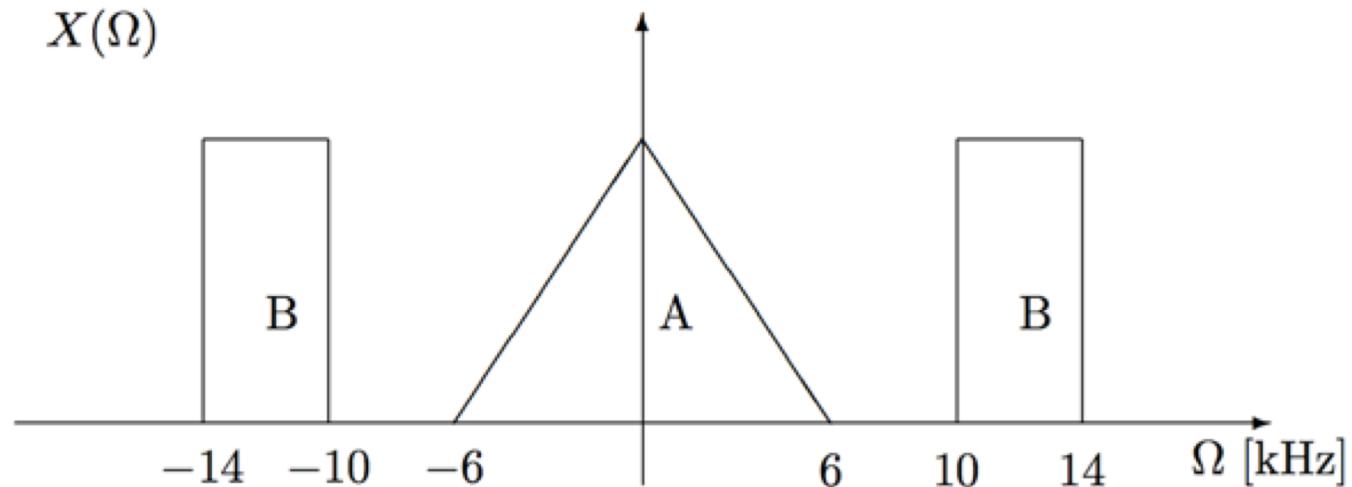


# Sampling solution 1a

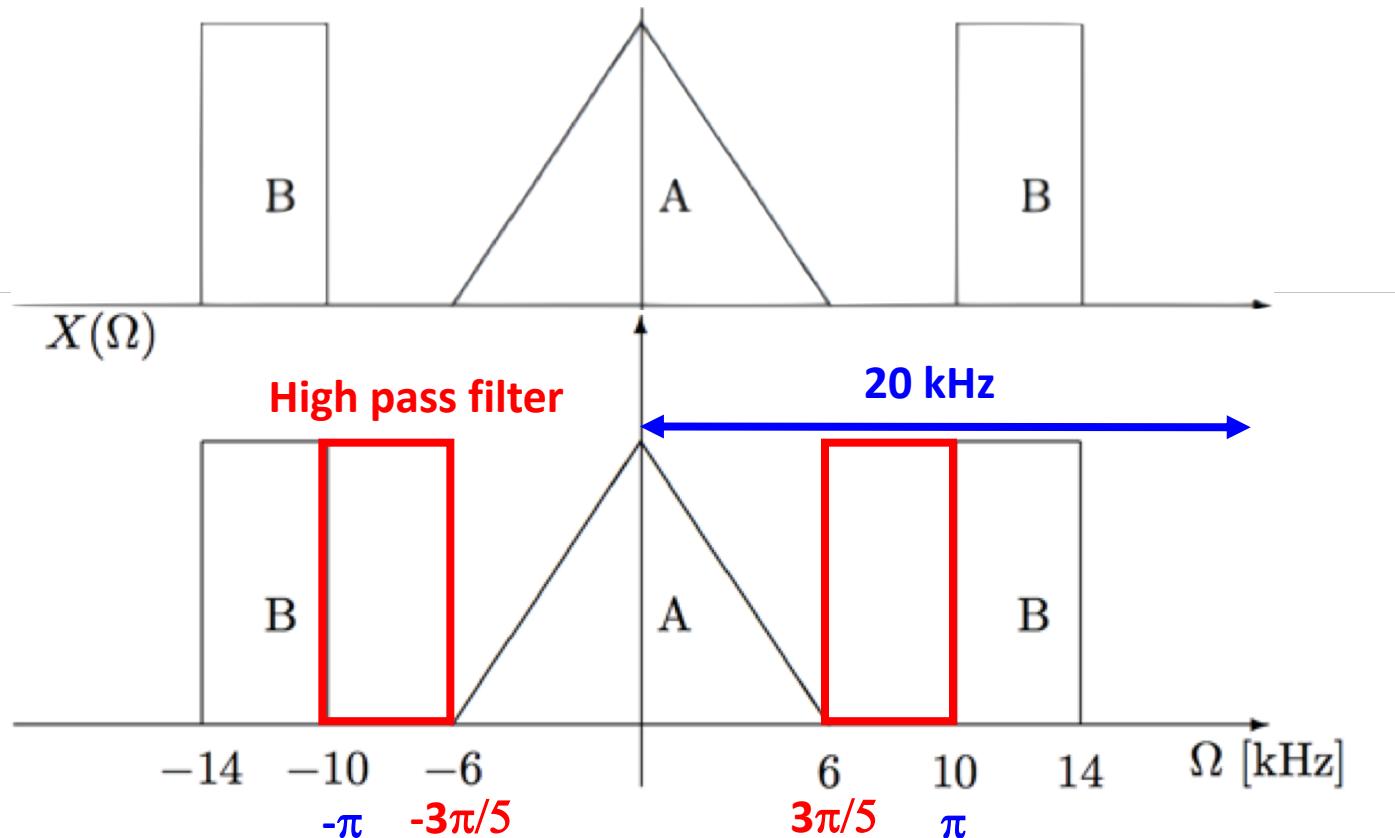


# Sampling question 1

4. An analog signal, whose spectrum is shown below, is to be processed with a digital filter using ideal C/D and D/C converters (with no analog anti-aliasing filters).
- a) (8 points) What is the minimum sufficient sampling rate to extract portion A of the signal? Sketch the magnitude of the digital filter that would be used at this sampling rate.
- b) (7 points) Repeat for portion B of the signal.

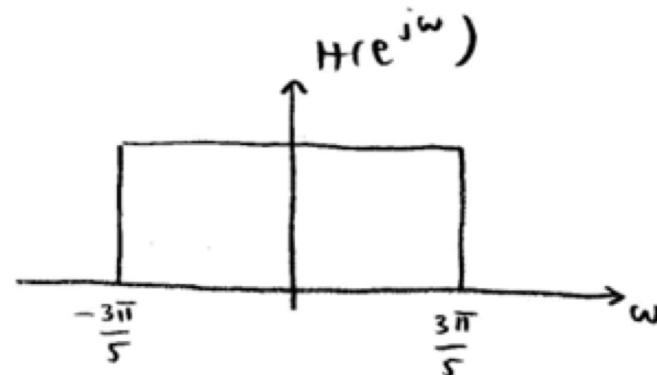
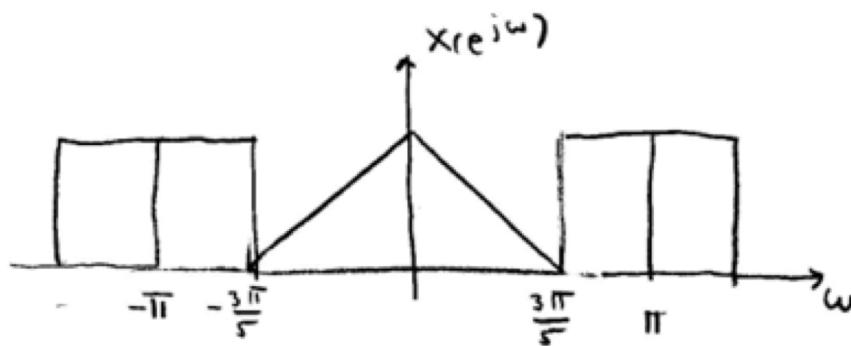


# Sampling solution 1b

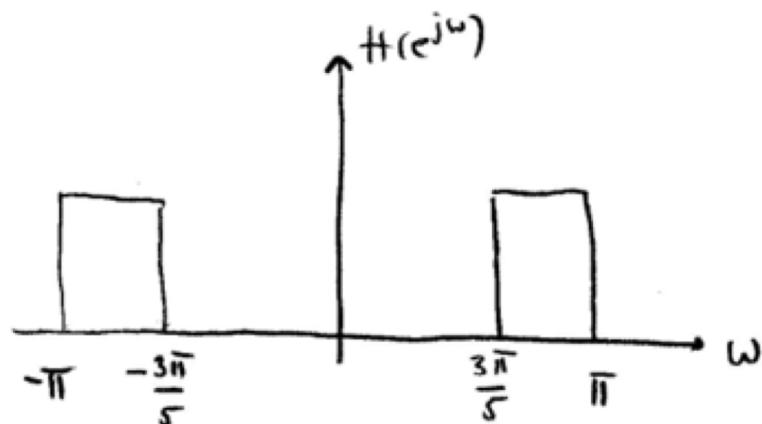
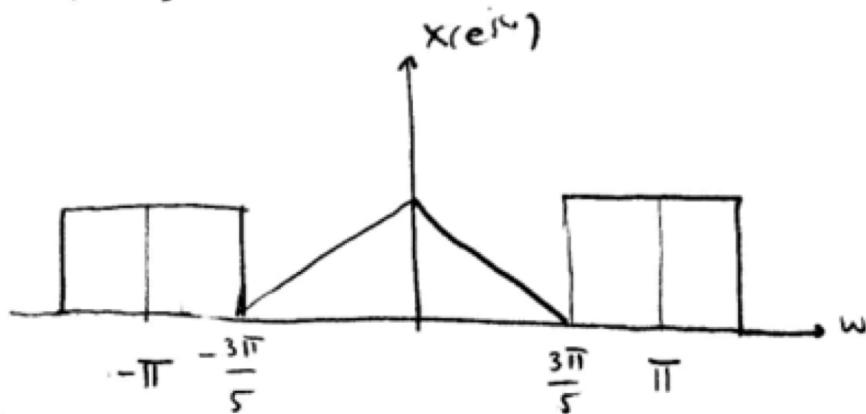


# Sampling solution 1

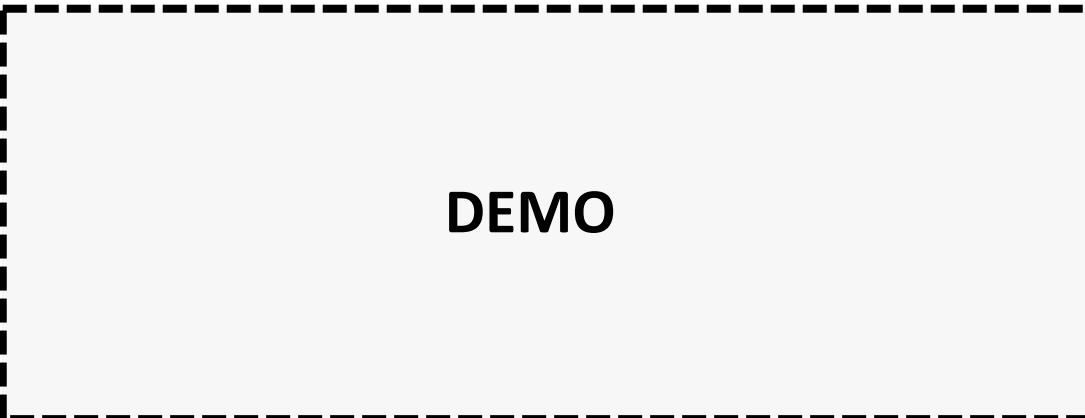
a)  $f_s = 20 \text{ kHz}$



b)  $f_s = 20 \text{ kHz}$



# Why do we care sampling



**DEMO**