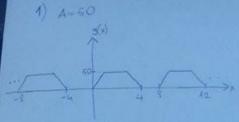
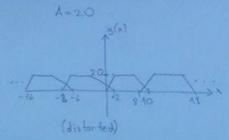


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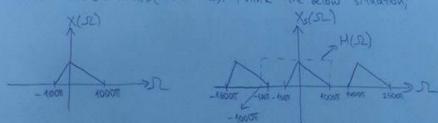


In above figures we can see that the rightmost edge of the signal closest to origin is at  $\frac{200}{A}$  and the leftmost edge of the signal shifted to its right is at x=8. For no overlap  $8>\frac{200}{A}$ . More generally,

## 1A1>25

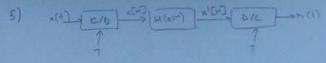
2) Minimum sampling frequency is two times the highest frequency component in the spectrum. In this case it must be 2000 rad/s or 1 kHz.

Since the spectrum is not symmetric there can be a misunderstanding that it is MOOT rad/s (550 Hz). Think the below situation;

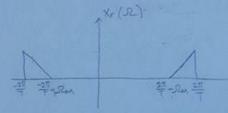


We have an X(2) signal that obeys the situation of the question and it is sampled with 750 Hz which is between 550 Hz and NkHz values. There is no aliasing after the sampling, But in order to recover the signal we must pass it through a lowpass filter at the last step, If we assume the filter H(JZ) has a real impulse response its spectrum must be symmetric. By that way it also passes a small portion of shifted signal which causes distortion. Hence minimum sampling frequency must be 1 kHz not 550 Hz

3) 0.25 was corresponds to 4 kHz or 800071 rad/s. Thus before the largest filtering step we have below signal. a) 0.75 ms => 1.33 EHz or 2660T red/s The 26607 rad/s only passes baseboard signal so we obtain the original signal with a scale factor of 6) 0.05 ms -20 kHz or 400007 radis. This filter passes backband signal and high frequency terms until 4000011 rads with a scale factor of We cannot recover the original signal in this care.







Equating 27 = 100001 = 7 = 0.2 ms

Also 27 -22 = 9000T > [2] = 1000T rad/s

Hence sampling period is T=0.2 ms. Our constraint is input signal must be boundlimited to 100011 rad/s or 500 Hz.

c) As can be seen in parts a and b 4(eim) and 42(eim) are same. Hence y, [n] and 42 [n] are the same

The answer does not change since there is no filtering order does not matter in this case.