

SIO 207A: Fundamentals of Digital Signal Processing

Class 12

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Decimation

- We aim to reduce the sampling rate by a factor of 8
- To avoid aliasing after decimation, we must pass the original signal through a low pass filter

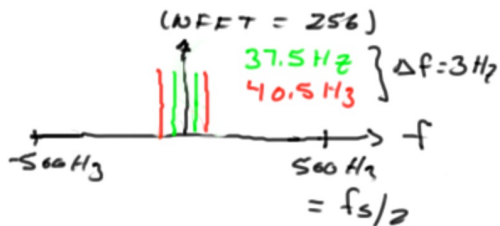


see also Sections 6.4 – 6.5 in Oppenheim & Schaffer, 1999: Decimation, Interpolation, Multirate Signal Processing

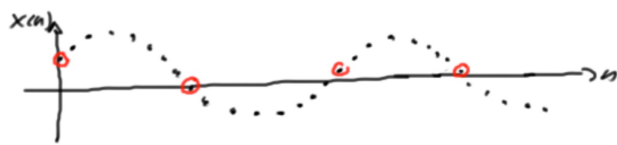
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Homework 5

- Resolve 2 closely spaced signals with a fixed length FFT



Decimation will lower the sampling rate

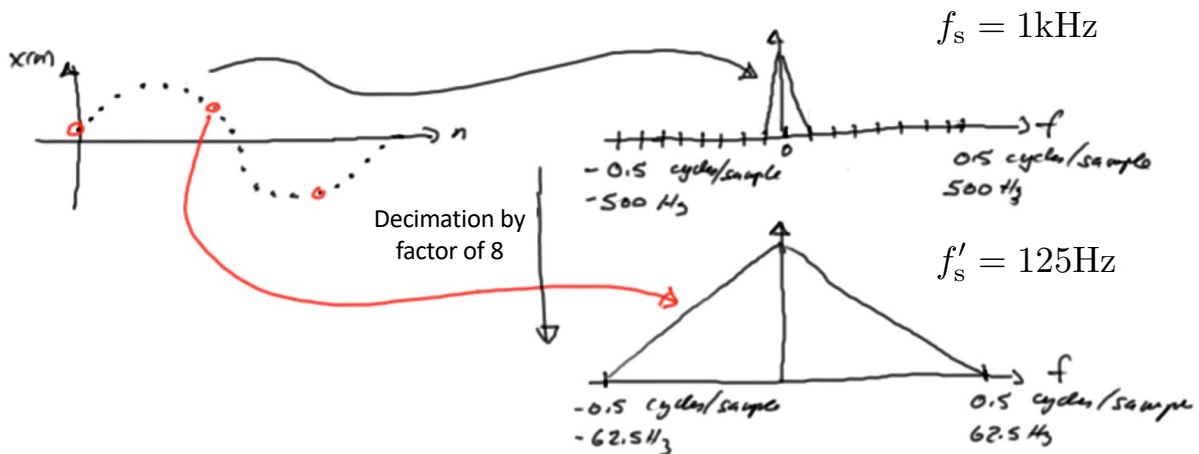


$$f_s = 1 \text{ kHz} \longrightarrow \text{FFT bin width: } f_s/256 \approx 4 \text{ Hz}$$

- Save one sample out of every 8 results in a decimation of factor 8
- This results in $f'_s = 125 \text{ Hz}$ and FFT bin width: $f'_s/256 \approx 0.5 \text{ Hz}$

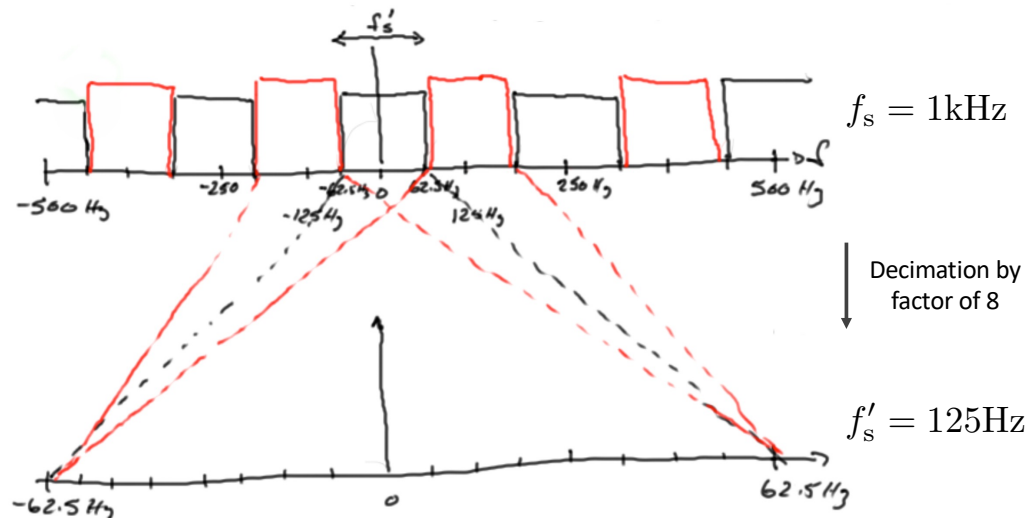
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Decimation – Mapping of Frequencies



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Mapping of Frequencies after Decimation and Aliasing Issue

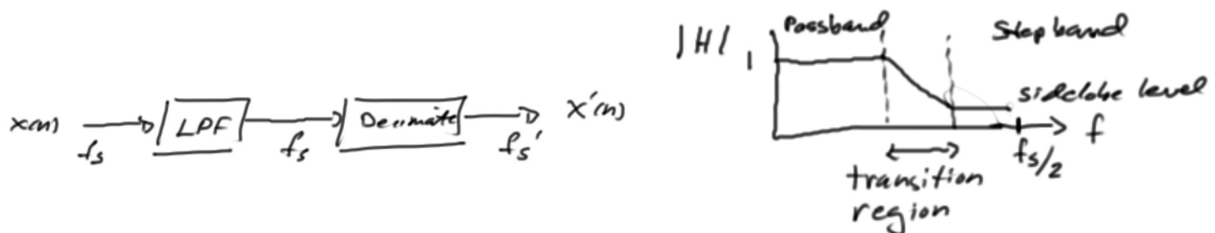


For example a signal at 260Hz in $x[n]$ will alias to 10Hz in $x'[n]$

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LP Filter to Avoid Aliasing

- To avoid aliasing after decimation, we must pass the original signal through a low pass filter
- This is essentially the same concept as an antialiasing filter prior to A/D conversion, i.e., we aim to remove all signal components that would alias into the signal band of interest



- The function of LPF is to attenuate high frequency content in $x[n]$ so aliasing is not a problem

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LP Filter Design in Matlab

- Design filter using `firpm` function in Matlab
 - pay particular attention to how frequencies are defined
 - note that there are distortions or ripples in both passband and stopband regions

Filter Specifications:

