## ECE478 Lab 7 Report

The Sampling Theorem

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## ABSTRACT:

The following lab explores the sampling theorem in communication theory. This is done via TIMS Hardware and MATLAB

## Experiment

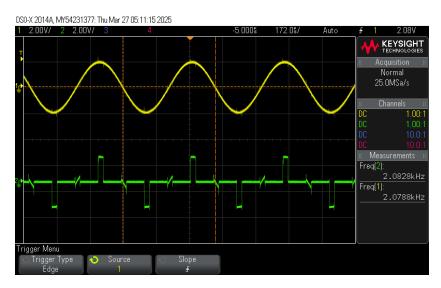
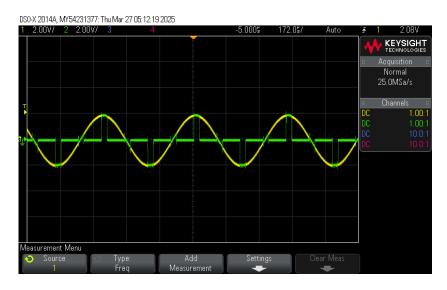


Figure 1: Message and sampled message.



 $Figure\ 2:\ Sampled\ message\ at\ reconstructable\ sample\ rate$ 

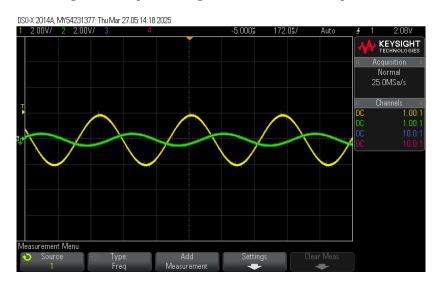


Figure 3: Message and reconstructed message post LPF

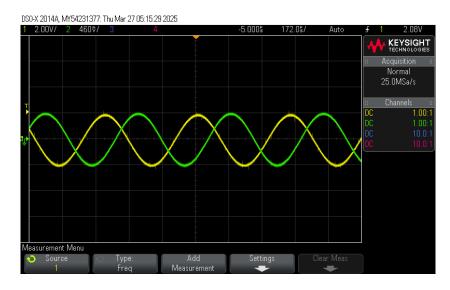
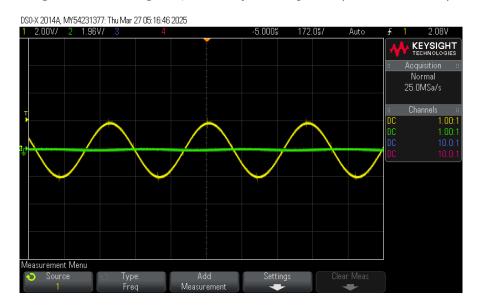


Figure 4: Same as Figure 3, with readjusted amplitude (shows attenuation)



 $Figure \ 5: Smallest \ sample \ width \ reconstruction$ 

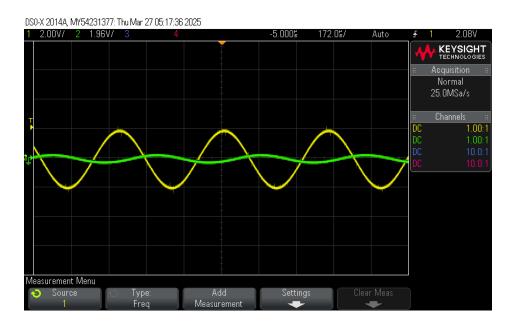


Figure 6: Moderate sample width reconstruction

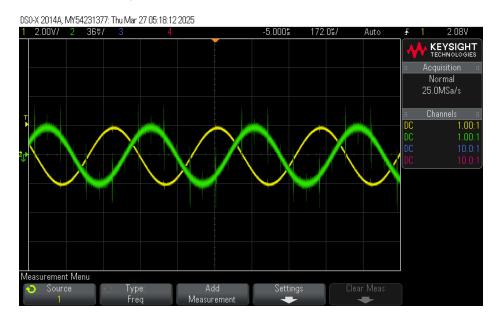


Figure 7: Largest sample width reconstruction

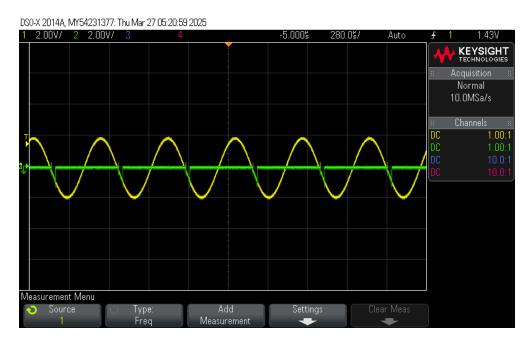


Figure 8: Synchronous

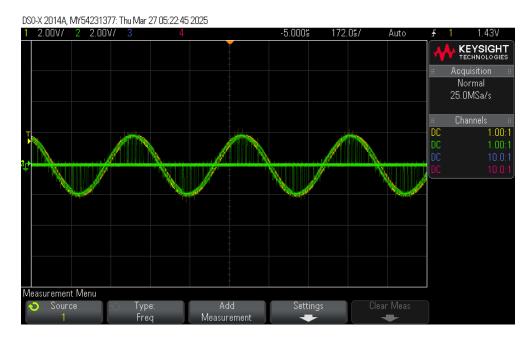


Figure 9: 2kHz message sampled with 8.333kHz TTL

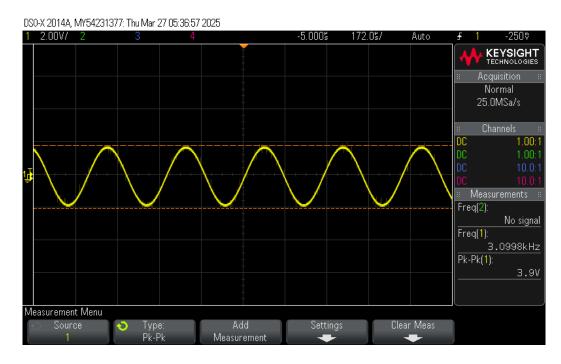
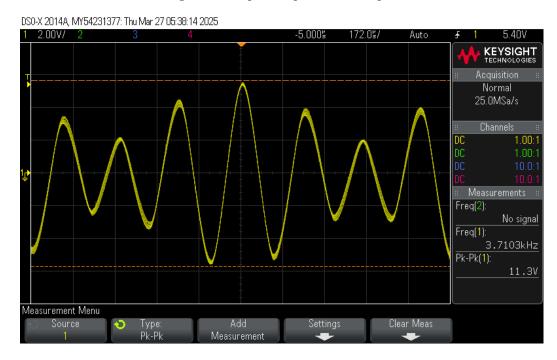


Figure 10: 4V peak to peak adder output



 $Figure \ 11: Adder \ output \ with \ maximum \ noise$ 

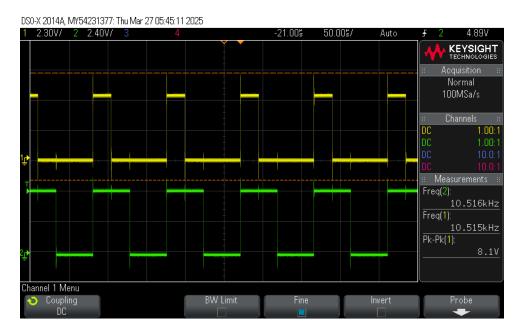


Figure 12: Synchronized, sample width 20% of sample period

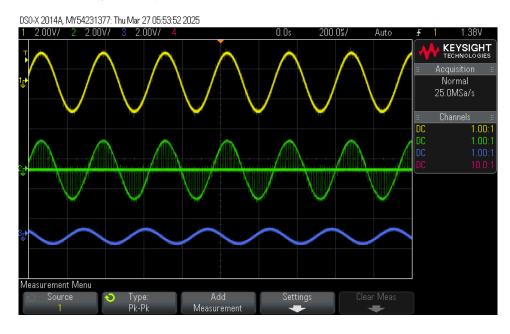


Figure 13: Message, sampled message, and reconstructed message

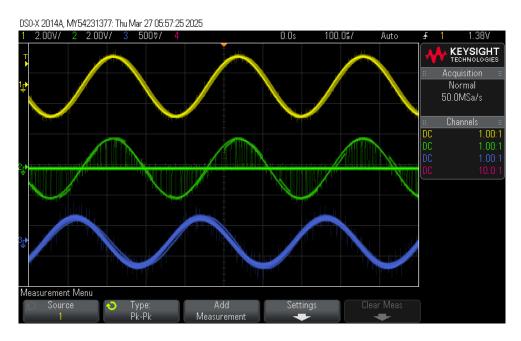


Figure 14: Reconstruction with f=7.087kHz, the minimum rate

The required sampling of the filter turns out to be slightly more than double the bandwidth of the signal, where we need 7.087kHz, while minimum by the sampling theorem for a 2kHz sinusoid would be 4kHz.

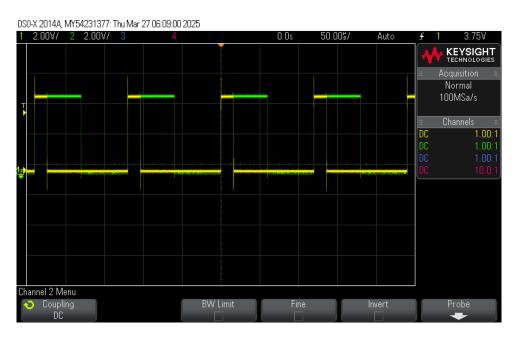


Figure 15: Sample duration 1/10 of clock period

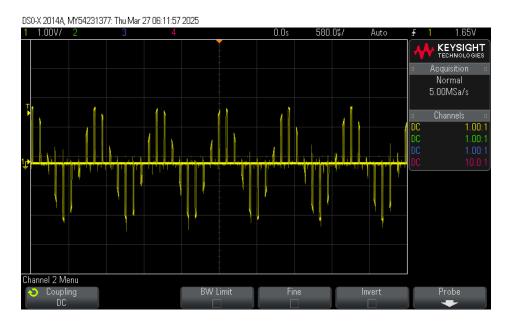


Figure 16: Sampled message

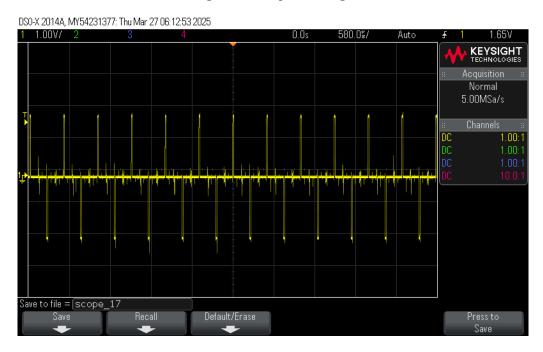


Figure 17: Message sampled with aliasing

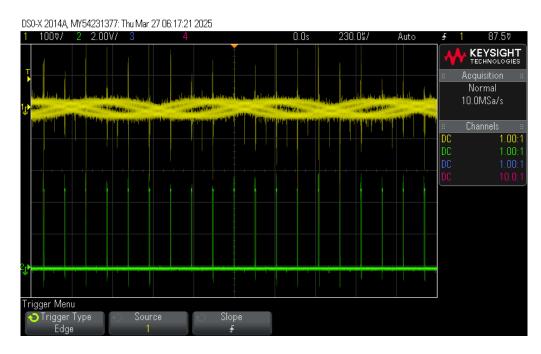


Figure 18: Reconstruction with closed LPF

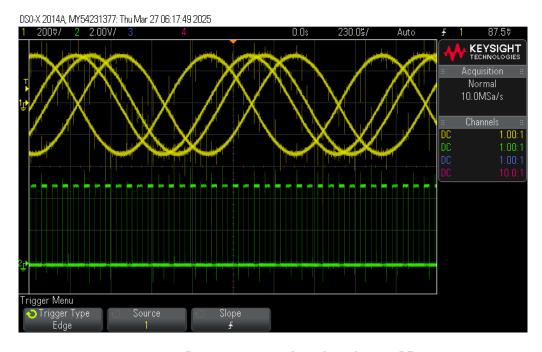


Figure 19: Reconstruction with moderately-open LPF

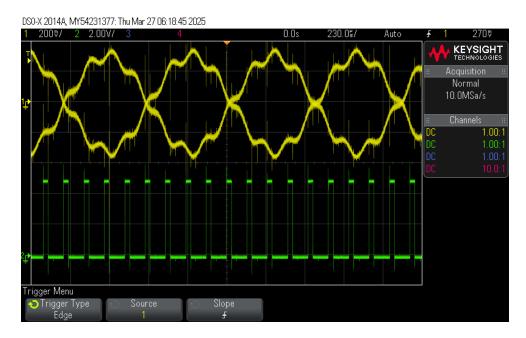
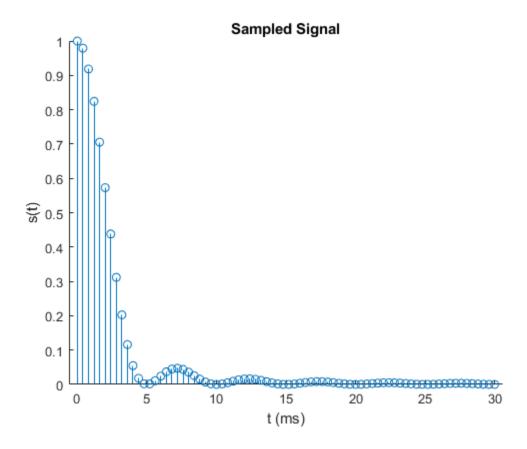


Figure 20: Reconstruction with fully-open LPF

## **MATLAB**



 $Figure\ 21:\ MATLAB\ sampled\ square\ sinc\ pulse$ 

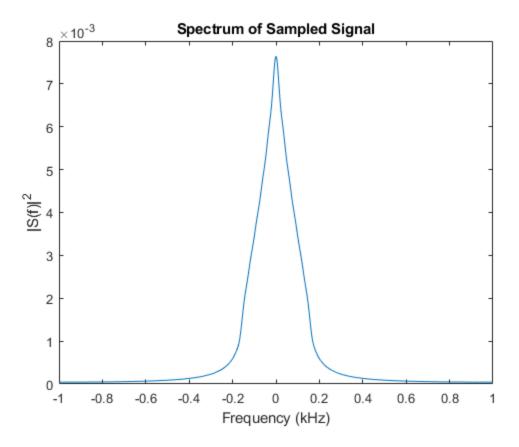


Figure 22: FFT of sampled square sinc pulse

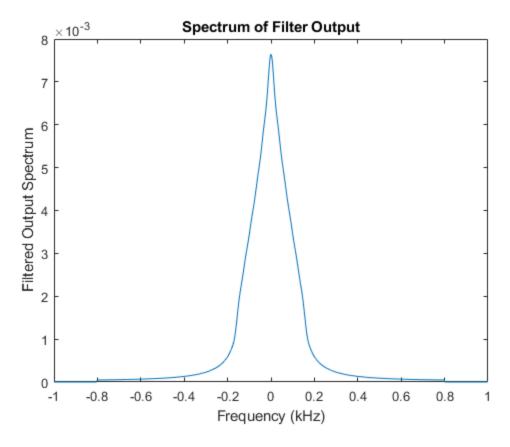


Figure 23: Output FFT of filtered sampled signal

Where we get the expected triangular frequency spectrum of a square sinc pulse as the LPF can pick out the original content of the sampled square sinc pulse.