

Your Paper

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February 10, 2020

Abstract

We studied signals.

1 Introduction and Background

1.1 The Fourier Transform

Where $V(t)$ is voltage as a function of time and $V(\nu)$ is voltage as a function of frequency ν .

$$V(\nu) = \int_{-T/2}^{T/2} V(t) \exp(2\pi i \nu t) dt$$

Where T is the length of the time sample.

$$V(t) = \frac{1}{\nu_s} \int_{-\nu_s/2}^{\nu_s/2} \tilde{V}(\nu) \exp(-2\pi i \nu t) d\nu$$

where ν_s is the sampling frequency used.

What happens if you change the bounds in the manner $(N / 2 - 1) / \nu_s$

1.2 Positive and Negative Frequencies

“complex inputs to a Fourier Transform break the positive/negative frequency degeneracy”

The sine function is odd:

$$\sin(-\omega t) = -\sin(\omega t)$$

The cosine function is even:

$$\cos(-\omega t) = \cos(\omega t)$$

We can generalize under Euler's identity:

$$A \exp(i\omega t) = A \cos(\omega t) + i \cdot A \sin(\omega t)$$

$$A \exp(-i\omega t) = A \cos(-\omega t) + i \cdot A \sin(-\omega t) = A \cos(\omega t) - i \cdot A \sin(\omega t)$$

1.3 Nyquist's Theorem and Aliasing

$$f_s > 2f_{max}$$

1.4 The Convolution and Correlation Theorems

The convolution of two functions $f(t)$ and $g(t)$ is defined as:

$$[f * g](\tau) = \int_{-T/2}^{T/2} f(t) g(\tau - t) dt$$

The correlation of two functions $f(t)$ and $g(t)$ is defined as

$$[f \star g](\tau) = \int_{-T/2}^{T/2} f(t) g(\tau + t) dt$$

You may want to prove this by substituting the Fourier transform

$$[f \tilde{*} g](\nu) \equiv \int_{-T/2}^{T/2} [f * g](\tau) \exp 2\pi i \tau \nu d\tau = \tilde{f}(\nu) \cdot \tilde{g}(\nu)$$

$$[f \star g](\nu) \equiv \int_{-T/2}^{T/2} [f \star g](\tau) \exp 2\pi i \tau \nu d\tau = \tilde{f}(\nu) \cdot \tilde{g}^*(\nu)$$

Now: how is the ACF relevant to this paper

1.5 Sideband Theory

All that sweet, sweet trig.

2 Methods

This is the equipment I used. These are the libraries and functions I used. This is how I used them (*vague* hint at your code). Uncertainty? Technical errors?

To begin our investigation of the Nyquist criterion, we decided on a sampling frequency $\nu_s = 6.25$ MHz at which the pico sampler (a PicoScope 2206A) remained for the duration of the data collection for all signals. We then used the N9310A RF Signal Generator to output signals at frequency ν_0 : fractions of the sampling frequency. For example, one of our input signals was $\nu_0 = .4\nu_s = 2.5$ MHz. For all signals, we performed visual inspections of the pico sampler's input by using a T-joint and connecting the output to an oscilloscope (a Rigol DS1052E). We used the ugradio module to perform data collection (`ugradio.pico.capture_data`) and, later on, to perform Fourier and inverse Fourier transforms (`ugradio.dft.dft` and `ugradio.dft.idft`).

To consider the frequency resolution for two signals of similar frequency, we employed a second signal generator. The 83712B has a lower limit of 10 MHz output frequency. Consequently, we set our original signal generator to small increments above that frequency and needed to increase the sampling frequency to $\nu_s = 31.25$ MHz.

To investigate the impact of noise on the Fourier transform, we switched the pico sampler's input to an NOD 5250 noise generator (6 MHz bandpass) at zero attenuation. We collected 32 blocks of 16000 samples from the PicoScope, this time using the sampling frequency $\nu_s = 62.5$ MHz.

We explored mixers and sideband theory with a Mini-Circuits ZAD-1. We the 83712B as the local oscillator and the N9310A as the RF input. We sent the output to the oscilloscope and PicoScope, this time using a 50Ω terminator to eliminate bounce-back interference. The ZAD-1 works best for signals above 10 MHz, so we set the local oscillator to $\nu_{LO} = 11$ MHz and collected two sets of data: one for which $\nu_{RF} = 11.55$ MHz and one for which $\nu_{RF} = 10.45$ MHz (thus, $\Delta\nu = .05\nu_{LO} = .55$ MHz).

The mixed signals were somewhat weak, so we used 1.5 dBm amplitudes on both signal generators. We again used a sampling frequency of $\nu_s = 62.5$ MHz, motivated by the following identities:

$$\sin(\nu_{LO}) \sin(\nu_{LO} + \Delta\nu) = \frac{1}{2}(\cos \Delta\nu - \cos(2\nu_{LO} + \Delta\nu)) \text{ by evenness of the cosine function}$$

and

$$\sin(\nu_{LO}) \sin(\nu_{LO} - \Delta\nu) = \frac{1}{2}(\cos \Delta\nu - \cos(2\nu_{LO} - \Delta\nu))$$

Now, if we expect to see two signals, $2\nu_{LO} + \Delta\nu = 23.1$ is the highest frequency that we would expect. Our PicoScope offers sampling rates by dividing the base 62.5 MHz. 31.25 MHz would have failed the Nyquist criterion, so we used the maximum sampling rate instead.

Finally, we constructed a single-sideband mixer by using two mixers. Both mixers used both the local oscillator and radio frequency signals, but one of the mixers' RF input was phase shifted by approximately $\pi/2$ by linking smaller cables together until we roughly satisfied the $\lambda/4$ criterion.

I don't remember the model of the other mixer

3 Notes to Self

What am I doing? Not ACF (5.3, 5.7) and procrastinating on 5.5 so that I can get some writing done.

I did ACF analysis for neither 5.3 nor 5.7

Major logic error: side-by-side captions are mutilated!

Idea: put signal and power spectrum plots side by side, not signals with signals

argue what the Nyquist criterion is, based on results.

* Include make and model of all equipment used.

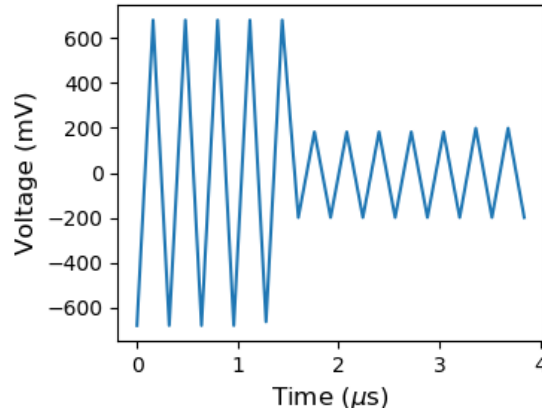


Figure 1: The oscilloscope displayed a constant signal throughout the period of data-taking. The data sampled from the pico sampler, however, reconstructs a signal with large aberrations in the first few microseconds.

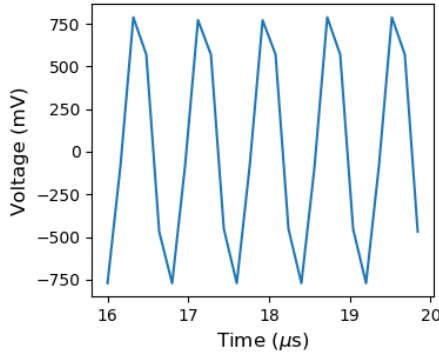


Figure 2: $\nu_0 = .2\nu_s = 1.25$ MHz.

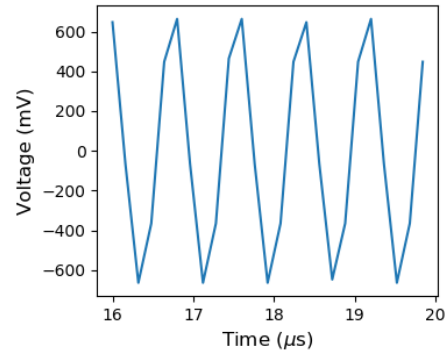


Figure 3: $\nu_0 = .8\nu_s = 5$ MHz.

* "Don't quote a number without the uncertainty and units."

Discussion on results for week 2, section 1

I need to include details about the equipment used, but how in-depth do I need to go? Current plan

* For most things, use model number and manufacturer

* When data analysis depends on a spec sheet, offer a brief summary of the specs to which you are referring to fine-tune your analysis

I need uncertainties on results but I do not yet know how to get these.

Closer to the end of writing, design a new subsection layout for the results section. You of course cannot make references like "5.2" and "7.3" in the submission.

4 Results

4.1 5.2

We took the default 16000 samples for each of our first signal samples. However, for the data analysis, we will be excluding the first 100 samples due to a peculiarity of the pico sampler which distorted these (see figure 1).

We may begin inspection of the samples with a qualitative approach. Figure 2 shows a signal which repeats about five times in the span of about 4 microseconds. This gives us 1.25 cycles per microsecond, or 1.25 MHz, as expected. Figure 3 still appears as a sine wave, but visual inspection

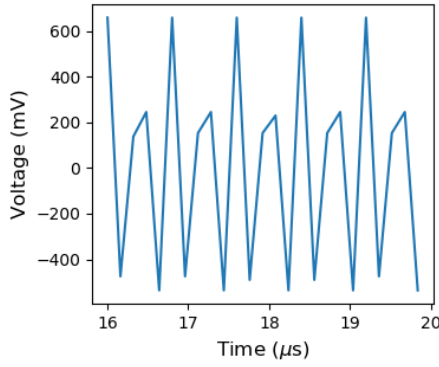


Figure 4: $\nu_0 = .4\nu_s = 2.5$ MHz.

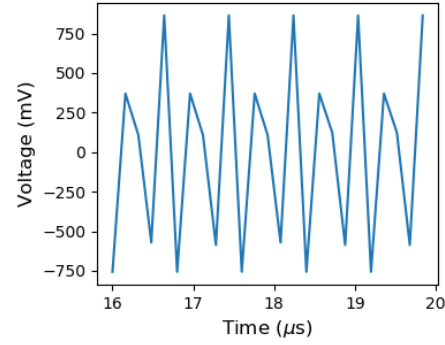


Figure 5: $\nu_0 = .6\nu_s = 3.75$ MHz.

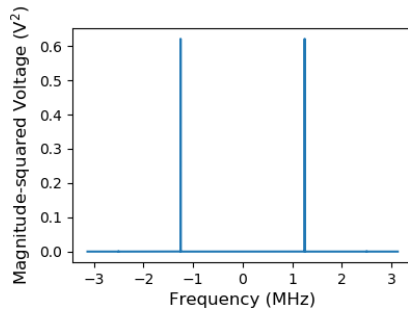


Figure 6: $\nu_0 = .2\nu_s = 1.25$ MHz. Peak amplitudes at ± 1.25 MHz

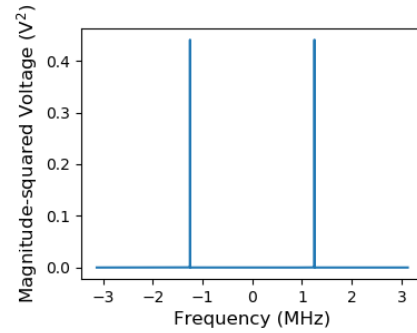


Figure 7: $\nu_0 = .8\nu_s = 5$ MHz. Peak amplitudes at ± 1.25 MHz

is no longer reliable. Again: we see about 5 samples in the span of about 4 microseconds. This would give 1.25 MHz, but because we know that our input signal was in fact $\nu_0 = 5$ MHz, we know that the waveform is aliased.

A more nuanced example comes from frequencies close to $.5\nu_s$: compare figures 4 and 9. Keeping in mind that only the frequency varied between trials and not the shape of the signal, a visual inspection is now confused: the sample looks like a sine wave in neither case. If take for granted that 4 is not a perfect sample, we may still visually estimate the frequency by noting 10 pairs of relative extrema (2.5 MHz).

may still say that there is a repeating signal with slightly under five repetitions in a span of 4 microseconds. This would give us slightly under 1.25 MHz, which is incorrect; thus we have qualitatively demonstrated an aliasing effect.

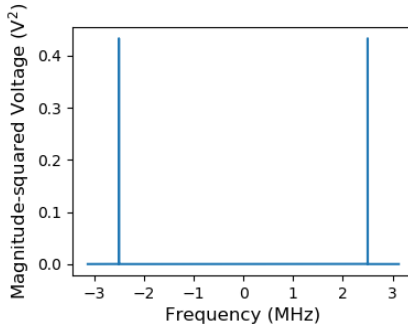


Figure 8: $\nu_0 = .4\nu_s = 2.5$ MHz. Peak amplitudes at ± 2.5 MHz

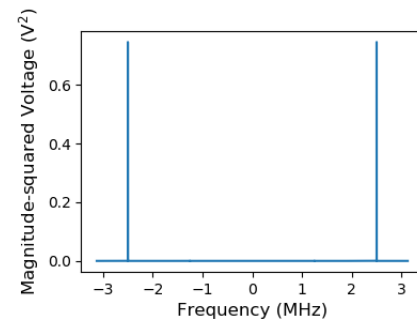


Figure 9: $\nu_0 = .6\nu_s = 3.75$ MHz. Peak amplitudes at ± 2.5 MHz

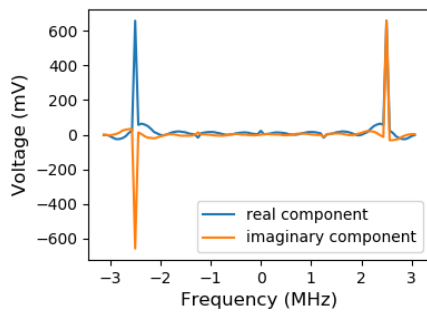


Figure 10: Voltage

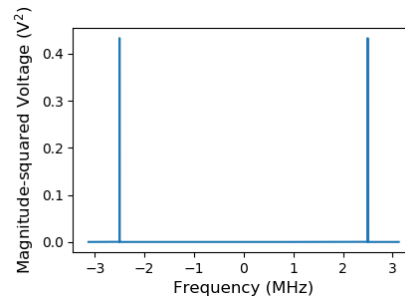


Figure 11: Powerish

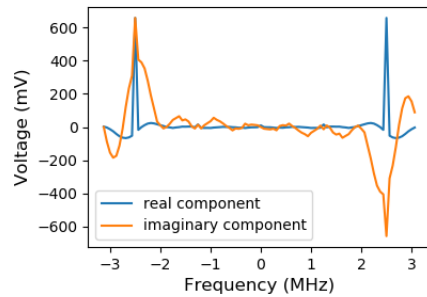


Figure 12: Voltage

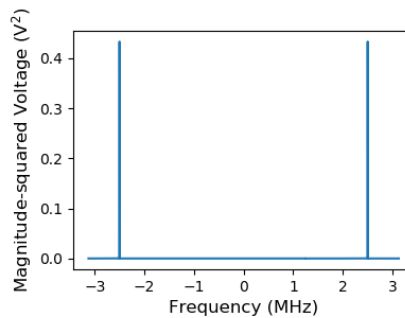


Figure 13: Powerish

4.2 5.3

“What does it mean, that the voltage spectra are complex? What do the real and imaginary parts represent? Is the imaginary part less ‘real’ than the real part? What does it mean, for frequencies to be negative versus positive?”

“Why might we use power spectra instead of voltage spectra, and vice versa?”

“According to the correlation theorem, the Fourier transform of the power spectrum should equal the ACF. Does it? Explain any differences.”

“you need to make sure `dft.idft` correctly infers the frequencies corresponding to each bins in your power spectrum array”

“When calculating a digital version of the correlation function, you have to worry about end effects. Suppose you are calculating an ACF for N samples with delays ΔN ranging up to $N/2$. Then the number of terms in the sum is always smaller than N because the delays spill over the edge of the available samples.”

How am I supposed to account for this?

Differences? Probably because I did not zero out the middle

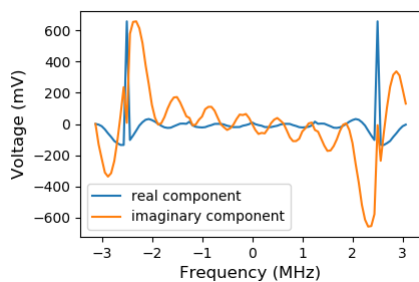


Figure 14: Voltage

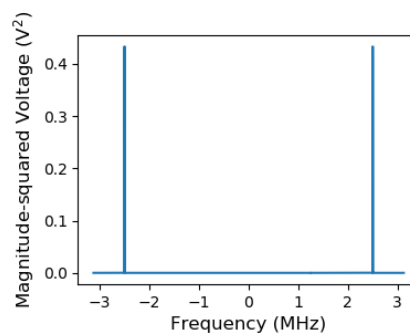


Figure 15: Powerish

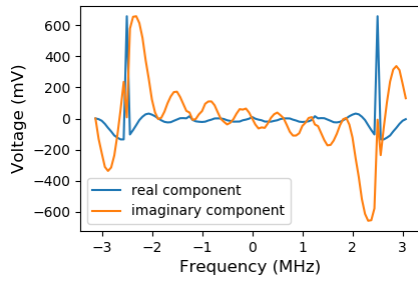


Figure 16: Voltage

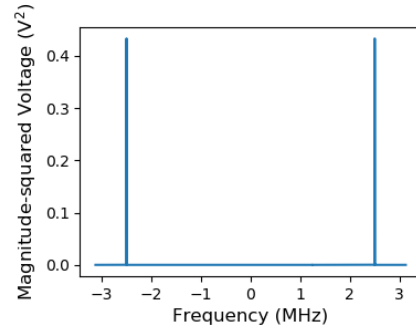


Figure 17: Powerish

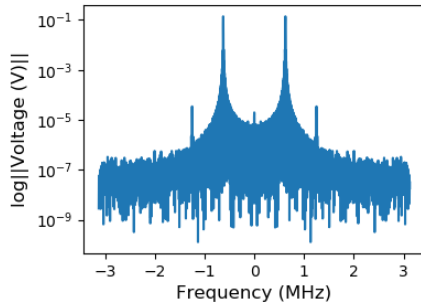


Figure 18: $\nu_0 = .1\nu_s = .625$ MHz

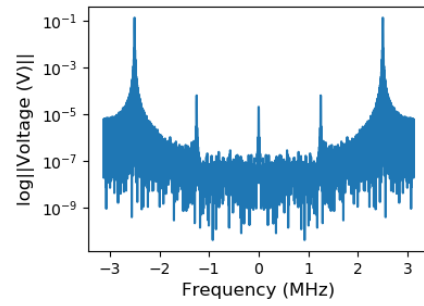


Figure 19: $\nu_0 = .4\nu_s = 2.5$ MHz

4.3 5.4

The y-axis is labeled incorrectly. We want log of square of voltage.

Recall that spectral leakage is introduced by the finite bounds on our Fourier transforms. Why does that math correspond to these results?

4.4 5.5

The separation looks pretty good so far.

4.5 5.6

4.6 5.7

First sample:

Mean = 3.899292452830189 mV Standard deviation = 20.079642416978363 mV Variance = 403.19203959371663 (mV)²

I am allowing the first 100 samples to contaminate. I do not know how much I should trod outside the 16000 recommendation.

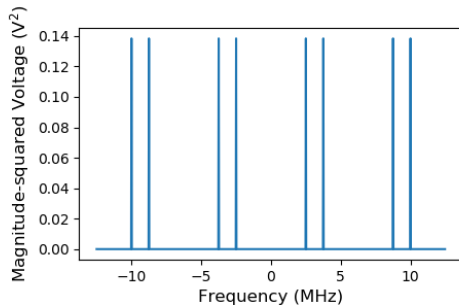


Figure 20: $\nu_0 = .4\nu_s = 2.5$ MHz. Fourth window.

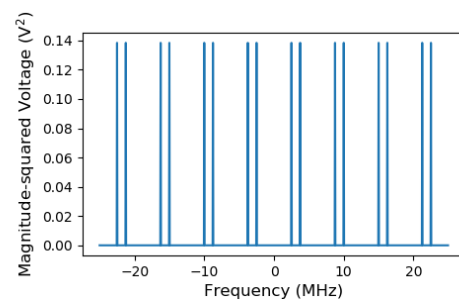


Figure 21: Same frequency. Eighth window.

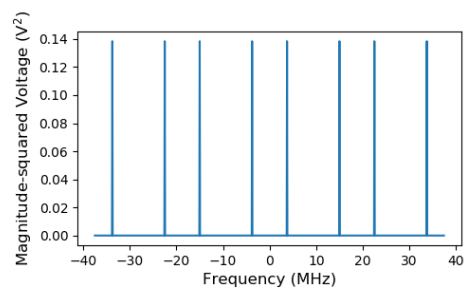


Figure 22: Same frequency. Twelfth window.

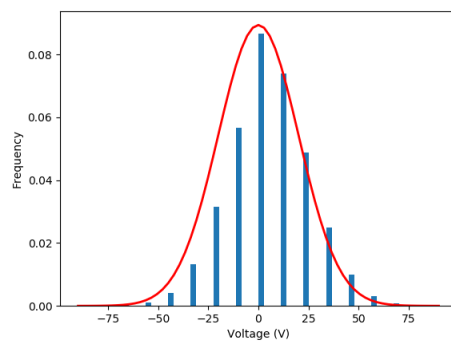


Figure 23

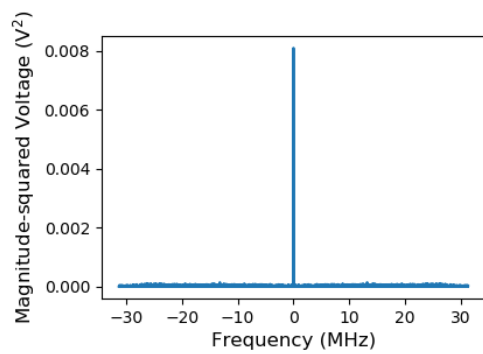


Figure 24

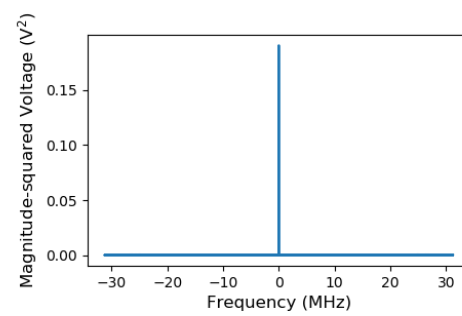


Figure 25

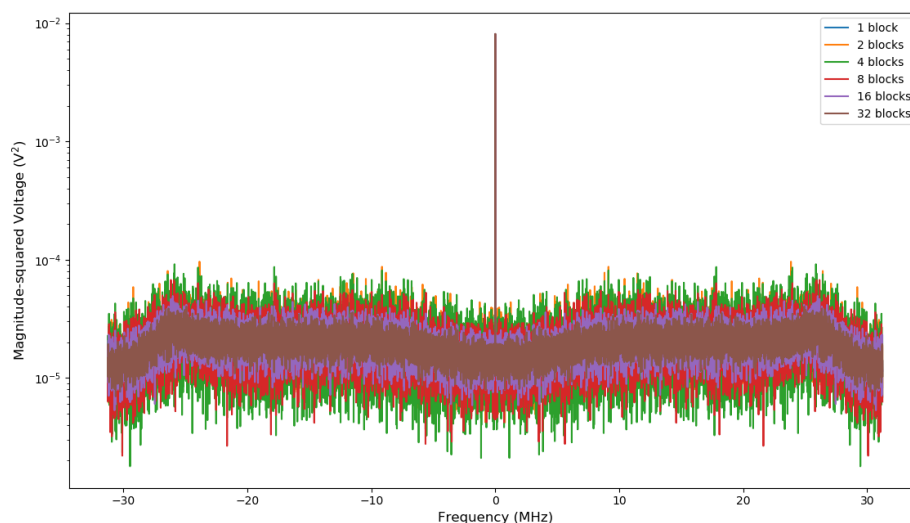


Figure 26

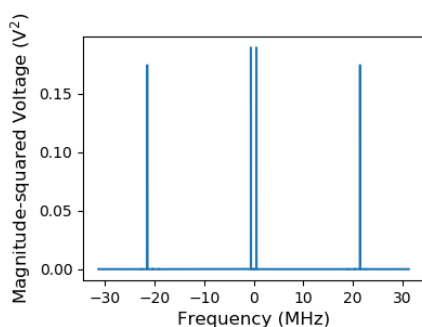


Figure 27

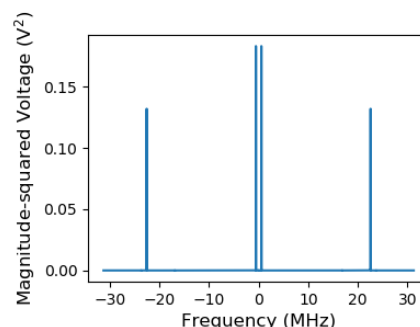


Figure 28

4.7 7.1

These figures are missing captions!

Why do the power spectra look the way they do. Upper sideband and lower sideband.

For the upper sideband, we can see spikes at almost the difference frequency (.575 MHz \approx .55 MHz). The other spikes are at 10.2 MHz? Why?

For the lower sideband, we see outer spikes at 9 MHz. The inner spikes are still at roughly the difference frequency...

4.8 7.2

4.9 7.3

Single side band mixer is more difficult to perform.

5 Conclusions

References

[Gre93] George D. Greenwade. The Comprehensive Tex Archive Network (CTAN). *TUGBoat*, 14(3):342–351, 1993.

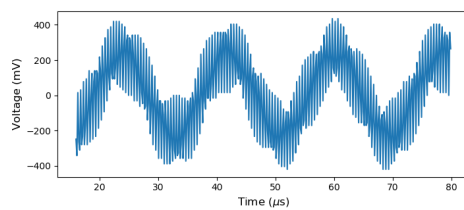


Figure 29