

DIGITAL SIGNAL PROCESSING: COSC390

Jordan Hanson

January 11, 2019

Whittier College Department of Physics and Astronomy

Previous lectures covered:

- Complex numbers 2: The Fourier series and Fourier transform (continuous and discrete)
- *Time-permitting*: The Laplace transform (continuous and discrete)

This lecture will cover: (Reading: **Chapter 2**)

- Statistics and probability: the normal distribution and other useful distributions
- Noise: digitization and sampling
- Noise: Spectral properties of noise, ADC and DAC

STATISTICS AND PROBABILITY: THE NORMAL DISTRIBUTION

The *mean*, μ , and *standard deviation*, σ , of a data set $\{x_i\}$ are defined as

$$\mu = \frac{1}{N} \sum_{i=1}^N x_i \quad (1)$$

$$\sigma^2 = \frac{1}{N-1} \sum_{i=1}^N (x_i - \mu)^2 \quad (2)$$

Octave commands:

```
x = randn(100,1);  
mean(x)  
std(x)
```

One nice theorem: *The variance is the average of the squares minus the square of the average.* Let $\langle x \rangle$ represent the average of the quantity or expression x . We have

$$\sigma_x^2 = \langle x^2 \rangle - \langle x \rangle^2 \quad (3)$$

Proof: observe on board.

STATISTICS AND PROBABILITY: THE NORMAL DISTRIBUTION

Note: There is a distinction between the *process or signal process* and the *the data*. Just because the data has a given μ and σ does not imply that the signal process has or will continue to have the exact same values of μ and σ . The underlying process could be *non-stationary*.

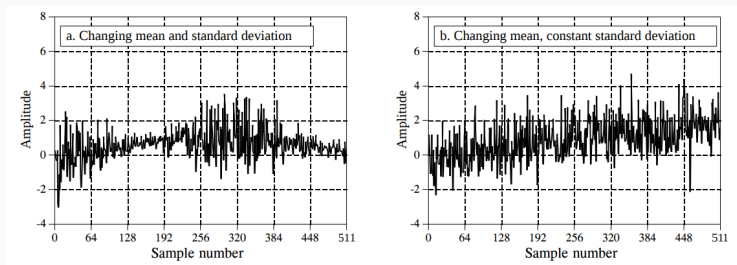


Figure 1: Signal processes in (a) and (b) are considered *non-stationary* because one or both of μ and σ depend on time.

A **histogram** is an object that represents the frequency¹ of particular values in a signal. For example, below is a histogram of 256,000 numbers drawn from a probability distribution:

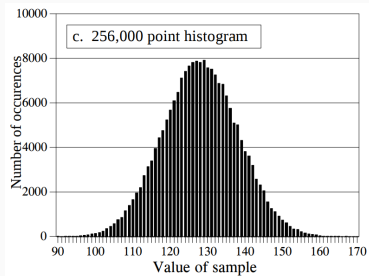


Figure 2: The histogram contains counts versus sample values.

¹Careful: the word frequency refers to the number of occurrences in the data, not a sinusoidal frequency.

The following octave code should reproduce something like Fig. 2 from the textbook:

```
x = randn(256000,1)*10.0+130.0;  
[b,a] = hist(x,100);  
plot(a,b,'o');
```

The function *randn*(*N*,*M*) draws $N \times M$ numbers from a normal distribution and returns them in the size the user desires. The function *hist*(*x*,*N*) creates *N* bins and sorts the data x_i into them.

For data that is appropriately stationary, we can use histograms to estimate μ and σ faster, since we only have to loop over bins rather than every data sample. Let H_i represent the counts in a given bin, and i represent the bin sample. We have:

$$\mu = \frac{1}{N} \sum_{i=1}^M i H_i \quad (4)$$

$$\sigma^2 = \frac{1}{N-1} \sum_{i=1}^M (i - \mu)^2 H_i \quad (5)$$

To obtain the mean in signal *amplitude*, you'll have to convert bin number to amplitude.

3.1
-0.03
1.2
0.2
-0.7
-1.45
2.2
-0.05
0.93
0.21

Table 1: Using Eq. 4 and 5, find estimates of μ and σ for this data.

```
x = [...];  
[b,a] = hist(x,4); %(How many bins?)
```

Some vocabulary:

- **normalization** - Total probability is 1.0. For pdf - the integral from $[-\infty, \infty]$ is 1.0. For pmf - the sum from $[-\infty, \infty]$ is 1.0.
- **pmf** - Probability mass function: A *normalized continuous function* that gives the probability of a value, given the value.
- **histogram** - Histograms are an attempted measurement of the pmf by breaking the data into discrete bins. Histograms can be *normalized* as well.
- **pdf** - Probability density function: A *normalized continuous function* that gives the probability density of a value, given the value. Integrating the *normalized* pdf between two values gives the probability of observing data between the given values.

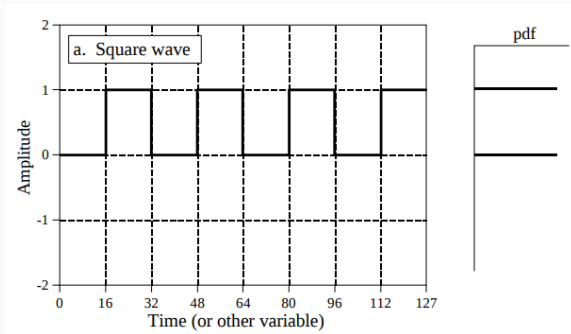


Figure 3: The square-wave signal spends equal time at 0.0 and 1.0, and the probability density function reflects that.

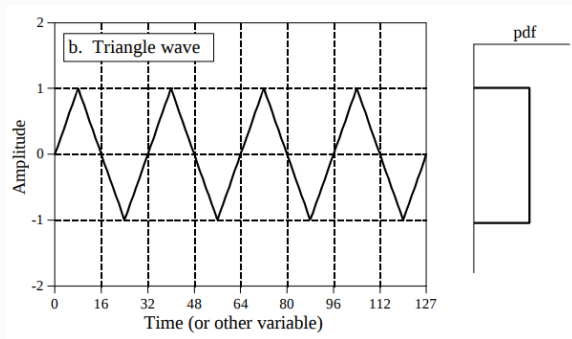


Figure 4: The triangle-wave signal spends equal time at all values *between 0.0 and 1.0*, and the probability density function reflects that.

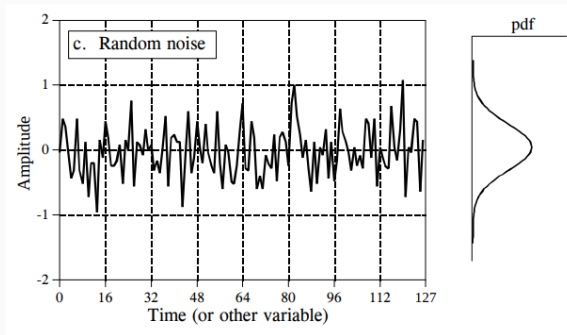


Figure 5: The random noise *usually* spends time near 0.0, but rarely it fluctuates to larger values.

NORMAL DISTRIBUTION

Normally distributed data decreases in probability at a rate that is proportional (1) to the *distance from the mean*, and that is proportional (2) to the *probability itself*.

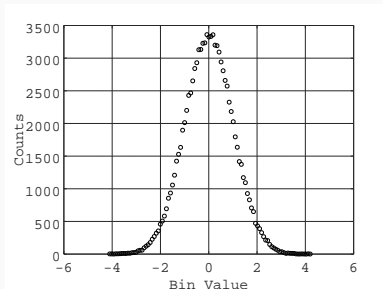


Figure 6: Normally distributed data counts decrease as measured further from the mean for *two reasons*.

Normal Distribution PDF

Let $p(x)$ be the PDF of normally distributed data x with mean μ . In order to obey conditions (1) and (2), the function $p(x)$ must be described by the following differential equation, where k is some constant.

$$\frac{dp}{dx} = -k(x - \mu)p(x) \quad (6)$$

Rearranging Eq. 6, we have

$$\frac{dp}{p} = -k(x - \mu)dx \quad (7)$$

Integrating both sides gives

$$\ln(p) = -\frac{1}{2}k(x - \mu)^2 + C_0 \quad (8)$$

Exponentiating,

$$p(x) = C_1 \exp\left(-\frac{1}{2}k(x - \mu)^2\right) \quad (9)$$

Ensuring that the PDF is *normalized* requires

$$\int_{-\infty}^{\infty} p(x)dx = 1 \quad (10)$$

But how do we integrate Eq. 9? First, a change of variables. Let $s = \sqrt{k/2}(x - \mu)$, so $ds = \sqrt{k/2}dx$. Then, we have

$$C_1 \sqrt{\frac{2}{k}} \int_{-\infty}^{\infty} \exp(-s^2) ds = 1 \quad (11)$$

Squaring both sides, we have

$$C_1^2 \frac{2}{k} \left(\int_{-\infty}^{\infty} \exp(-s^2) ds \right)^2 = 1 \quad (12)$$

Let's pretend the two factors of the integral involve different variables:

$$C_1^2 \frac{2}{k} \left(\int_{-\infty}^{\infty} \exp(-x^2) dx \right) \left(\int_{-\infty}^{\infty} \exp(-y^2) dy \right) = 1 \quad (13)$$

Now we have

$$C_1^2 \frac{2}{k} \int_{-\infty}^{\infty} \exp(-(x^2 + y^2)) dx dy = 1 \quad (14)$$

Change to polar coordinates ($x^2 + y^2 = r^2$)

$$C_1^2 \frac{2}{k} \int_0^{\infty} \int_0^{2\pi} r \exp(-r^2) dr d\phi = 1 \quad (15)$$

NORMAL DISTRIBUTION

One more substitution: $u = r^2$, and $du = 2rdr$:

$$-\frac{C_1^2}{k} \int_0^\infty \int_0^{2\pi} \exp(-u) du d\phi = 1 \quad (16)$$

Solving for C_1 , we find

$$C_1 = \sqrt{\frac{k}{2\pi}} \quad (17)$$

Thus the pdf of normally distributed data is

$$p(x) = \sqrt{\frac{k}{2\pi}} \exp\left(-\frac{1}{2}k(x - \mu)^2\right) \quad (18)$$

Let's defined $k = \frac{1}{\sigma_x^2}$ so that it's clear the exponent has the proper ratio of units:

$$\boxed{p(x) = \sqrt{\frac{1}{2\pi\sigma_x^2}} \exp\left(-\frac{1}{2}\left(\frac{x - \mu}{\sigma_s}\right)^2\right)} \quad (19)$$

STATISTICS AND PROBABILITY: PROGRAMMING WITH OCTAVE

More on the *hist* function in octave²

```
pkg install -forge io
pkg install -forge statistics
pkg load statistics
pkg help histfit
histfit(randn(1000,1))
histfit(rand(1000,1))
```

Let's work out the σ of a *flat* distribution between $[0, 1]$. What is it for a flat distribution between $[-1, 1]$? (We can derive this by hand as well if we cannot access statistics package).

²I hope this works, but if not, it's ok.

Some interesting notation for normal distributions:

$$N(\mu, \sigma) = \sqrt{\frac{1}{2\pi\sigma^2}} \exp\left(-\frac{1}{2}\left(\frac{x-\mu}{\sigma}\right)^2\right) \quad (20)$$

Let's write a function **NGaus.m** that produces the Gaussian probability given μ and σ :

```
function ret = NGaus(mu,sigma,x)
    ...
endfunction
```

Now let's write a function *NRand* that sums N uniformly-distributed (flat) random variables x :

```
function ret = NRand(n)
    ret = sum(rand(n,1));
endfunction
```

Create a histogram of a few hundred outputs of *NRand*. What do you notice about the pmf? Let's plot *NGaus* on the same axes as the histogram of *NRand*. How do they compare?

We are on our way to producing $N(0,1)$ distributed numbers, and therefore our first **noise** signals...

The Box-Muller method for $N(0, 1)$ distruted numbers:

$$X_1 = \sqrt{-2 \ln(U)} \cos(2\pi V) \quad (21)$$

$$X_2 = \sqrt{-2 \ln(U)} \sin(2\pi V) \quad (22)$$

Try this in octave... More vocabulary:

- **cdf** - Cumulative distribution function: Probability that a continuous random variable X is less than some value x . For a given pdf, the cdf $\Phi(X)$ is the integral of the total probability on $[-\infty, x]$. The derivative of the pdf is related to the pdf via the fundamental theorem of calculus.

If the pdf follows $f(x)$, then

$$\Phi(X \leq x) = \int_{-\infty}^x f(x) dx \quad (23)$$

The cdf of $N(0, 1)$ has an expected shape, but can't be expressed with elementary functions.

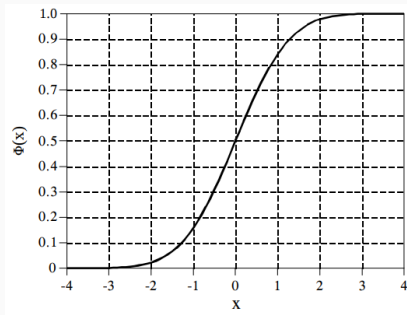


Figure 7: The cumulative distribution of the normal distribution. Although we can plot it, it's hard to write. We will discuss the *erf* and *erfc* functions in the near future.

STATISTICS AND PROBABILITY: OTHER USEFUL DISTRIBUTIONS

We now know how to obtain random uniform numbers (**rand**) in octave, and have algorithms (Box-Muller) and functions (**randn**) in octave for $N(0, 1)$ ³. What if we require a *different pdf*? One technique is to use *inverse transform sampling*:

1. For the pdf $p(x)$, work out the cdf $\Phi(x)$.
2. Generate a sample of uniform random numbers $u_i \in [0, 1]$.
3. Call $\Phi(u_i)$.

Write an octave script that generates exponentially-distributed numbers. Be careful to normalize when comparing to the expected pdf $\propto \exp(-x)$.

³This can be scaled to any μ and σ values we need.

Octave has many pre-programmed distributions. Although system noise is usually normally distributed, it's good to know some of these:

<https://octave.org/doc/v4.2.0/Distributions.html>

NOISE: DIGITIZATION AND SAMPLING

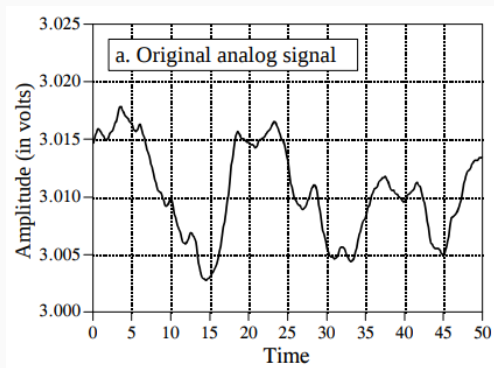


Figure 8: An example of analogue data from chapter 2 of the text. Both the dependent and independent axes are continuous.

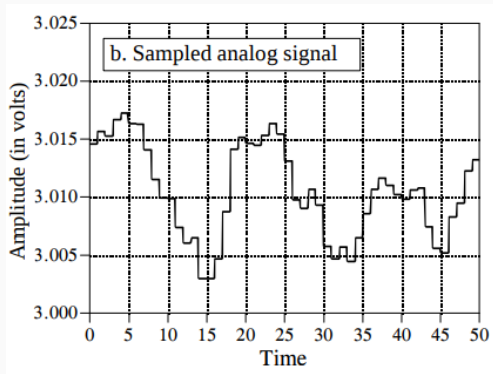


Figure 9: The same signal from Fig. 8, except a *sample-and-hold* action has been applied to the independent variable.

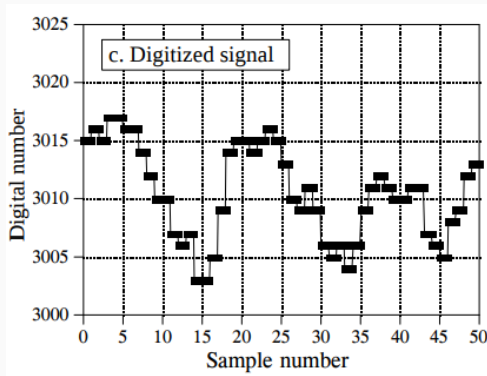


Figure 10: The same signal from Fig. 9, except a *digitization* action has been applied to the dependent variable.

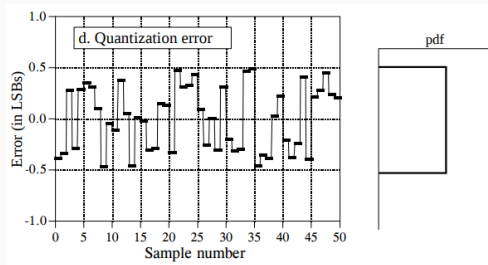


Figure 11: The error incurred by the *digitization* action from Fig. 10. The y-axis is expressed in units of LSB (least significant bit - more in a second). Turns out we know the σ of this error: $LSB/\sqrt{12}$.

A model for a particular value in Fig. 9, the sample-and-hold action⁴ is

$$s_n(t) = f(n\Delta t)\text{square}(t - n\Delta t) \quad (24)$$

where $f(n\Delta t)$ is the function or data value, and

$$\text{square}(t) = 1, \quad |t| \leq T/2 \quad (25)$$

The entire N -sample data-set or signal is

$$s(t) = \sum_{n=0}^{N-1} f(n\Delta t)\text{square}(t - n\Delta t) \quad (26)$$

⁴Technically, this is the 0th-order hold, and there are other (much) less common choices.

Sample/Hold Signal Model

$$s(t) = \sum_{n=0}^{N-1} \text{square}(t - n\Delta t) \quad (27)$$

Several important questions:

1. What is $S(\omega)$?
2. What are the important relationships between Δt , N , and the frequencies present in the data?
3. How precisely does $s(t)$ represent the data?

The Fourier transform of $s(t)$ may be obtained using a combination of properties of the Fourier transform, plus the result obtained for $F\{\text{square}(t)\}$. Let $x = \omega\Delta t/2$. The result is (observe on board):

$$S(\omega) = \text{sync}(x) \sum_{n=0}^{N-1} f(n\Delta t) \exp(-j\omega n\Delta t) \Delta t \quad (28)$$

The factor at right is a discrete version of the Fourier Transform. Let the DFT represent the discrete Fourier transform on the right. Equation 28 may be written

$$S(\omega) = \text{DFT}\{f(t)\} \text{sync}(x) \quad (29)$$

The spectrum of a sampled signal is the **convolution** of the discrete Fourier transform of the signal and the sync function with a period of the time between samples.

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

$$(f \circ g)(t) = \int_{-\infty}^{\infty} f(\tau)g(t - \tau)d\tau \quad (30)$$

Convolution theorem: The Fourier transform of the convolution of two functions $f \circ g$ is

$$F\{f \circ g\} = F(\omega)G(\omega) \quad (31)$$

NOISE: DIGITIZATION AND SAMPLING

The DFT can contain only have a finite number of frequencies, since it is discrete. What are the limits of this?

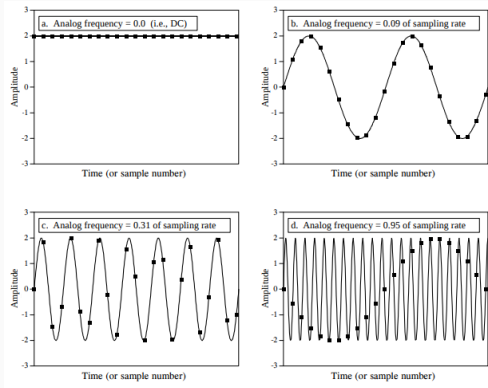


Figure 12: Various degrees of sampling.

Notice that the sinc function has a zero, which occurs at $x = \pi$, for some frequency f_s . This implies that

$$\pi = \frac{\omega \Delta t}{2} \quad (32)$$

$$\pi = \frac{2\pi f_s \Delta t}{2} \quad (33)$$

$$f_s = \frac{1}{\Delta t} \quad (34)$$

The frequency f_s is known as the *sampling frequency*.

We have finally arrived at the *sampling theorem*:

Sampling Theorem

A signal containing frequencies less than or equal to $f_{crit} = f_s/2$ can be perfectly reconstructed.

Let's go back and think about Fig. 12.

NOISE: DIGITIZATION AND SAMPLING

The DFT can contain only have a finite number of frequencies, since it is discrete. What are the limits of this?

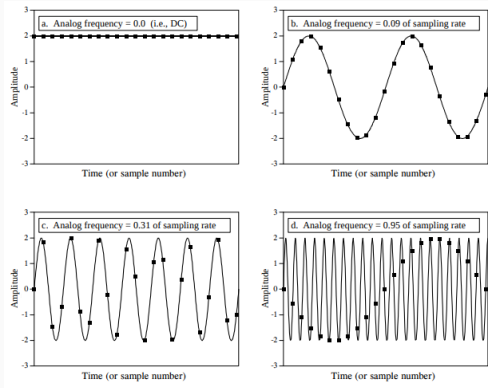


Figure 13: What if the sine wave had a frequency of f_c ?

CONCLUSION

Previous lectures covered:

- Complex numbers 2: The Fourier series and Fourier transform (continuous and discrete)
- *Time-permitting*: The Laplace transform (continuous and discrete)

This lecture will cover: (Reading: **Chapter 2**)

- Statistics and probability: the normal distribution and other useful distributions
- Noise: digitization and sampling
- Noise: Spectral properties of noise, ADC and DAC