

# Regression Analysis

5 Oct 2012

## (I) PROBLEM

The problem is as follows: Given a set of  $N$  data points

$$\mathcal{D} = \{(x_1, y_1), (x_2, y_2), \dots, (x_N, y_N)\},$$

find a function that best approximates the underlying function  $f(x_i) = y_i$  which relates the data. The nature of the underlying function that defines the data or the details of how the data was generated is usually unknown. These data points might be totally random or may be related by a characteristic function.

## (II) THEORY

In the absence of the underlying function, one could treat the data as a linear combination of some set of functions.

$$\begin{aligned}\hat{y} &= \sum_{i=1}^M w_i \phi_i(x) \\ &= \bar{w}^T \bar{\phi}(x),\end{aligned}\tag{1}$$

where  $M$  is the number of terms,  $\bar{w} = [w_1 w_2 \dots w_M]^T$  (superscript  $T$  refers to the matrix transpose), and  $\bar{\phi}(x) = [\phi_1(x) \phi_2(x) \dots \phi_M(x)]^T$ . Here,  $w_i$  is the weight corresponding to the function  $\phi_i$ .

### (II.I) ORTHOGONAL FUNCTIONS

Consider two functions  $\phi_1(x)$  and  $\phi_2(x)$  defined over the range  $[a, b]$ . The *inner product* of these functions is defined as

$$\langle \phi_1, \phi_2 \rangle = \int_a^b \phi_1(x) \phi_2(x)$$

$\phi_1$  and  $\phi_2$  are said to be *orthogonal* if  $\langle \phi_1, \phi_2 \rangle = 0$ . If there is a set of functions  $\{\phi_1, \phi_2, \dots, \phi_M\}$  defined over a range, then these functions form an orthogonal set if the inner product of any pair of these orthogonal functions  $\langle \phi_i, \phi_j \rangle = 0$  for  $i \neq j$ .

### (II.II) FOURIER SERIES

A Fourier series is a decomposition of a periodic function into a sum of

infinite sine/cosine functions. Any periodic function  $f(x)$  with fundamental period  $T$  can be represented using Fourier series as:

$$f(x) = a_0 + \sum_{n=1}^{\infty} \left( a_n \cos \frac{2n\pi x}{T} + b_n \sin \frac{2n\pi x}{T} \right)$$

The Fourier coefficients  $a_n$  and  $b_n$  can be determined from the following integrals:

$$\begin{aligned} a_0 &= \frac{1}{T} \int_0^T f(x) dx \\ a_n &= \frac{2}{T} \int_0^T f(x) \cos \frac{2n\pi x}{T} dx \\ b_n &= \frac{2}{T} \int_0^T f(x) \sin \frac{2n\pi x}{T} dx \end{aligned}$$

### Fourier series representation of a Sawtooth wave

Consider the sawtooth function below with a peak value of 1. The func-

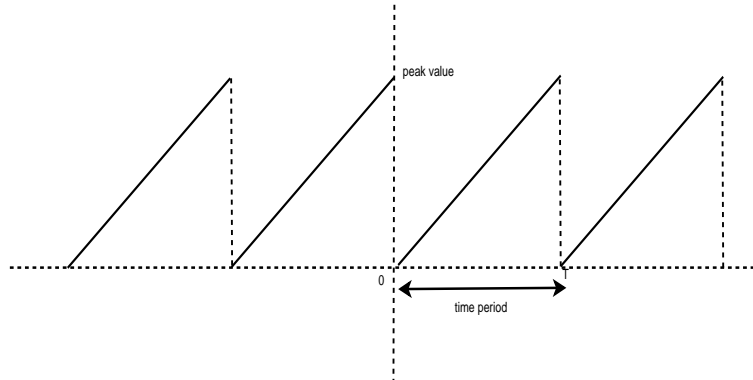


Figure 1: Sawtooth function

tional form of this wave is

$$f(x) = \frac{x}{T}$$

The sawtooth wave is a periodic function (time period  $T$ ) and hence can be decomposed using Fourier series representation. The components of the Fourier series are therefore given by:

$$\begin{aligned} a_0 &= \frac{1}{T} \int_0^T \frac{x}{T} dx \\ &= \frac{1}{2} \\ a_n &= \frac{2}{T} \int_0^T \frac{x}{T} \cos \left( \frac{2n\pi x}{T} \right) dx \\ &= 0 \end{aligned}$$

$$\begin{aligned}
b_n &= \frac{2}{T} \int_0^T \frac{x}{T} \sin\left(\frac{2n\pi x}{T}\right) dx \\
&= -\frac{1}{n\pi}
\end{aligned}$$

The Fourier series is therefore given by

$$f(x) = \frac{1}{2} - \frac{1}{\pi} \sum_{n=1}^{\infty} \frac{1}{n} \sin\left(\frac{2n\pi x}{T}\right) \quad (2)$$

### Fourier series representation of a Square wave

Consider the square wave shown below with a peak value of 1.

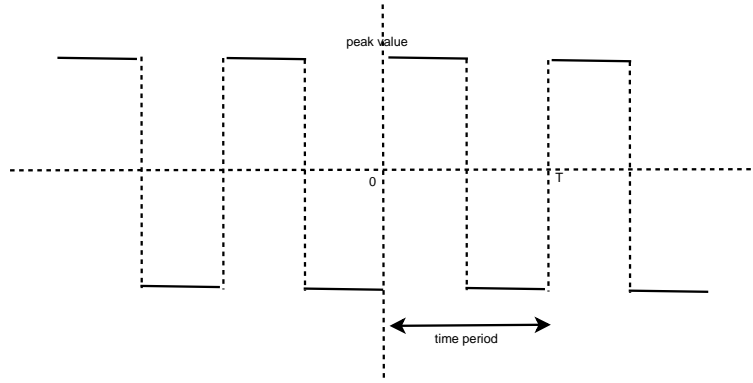


Figure 2: Square function

The functional form of this wave is

$$f(x) = \begin{cases} +1 & \text{if } 0 < x < \frac{T}{2}, \\ -1 & \text{if } \frac{T}{2} < x < T. \end{cases}$$

The components of the Fourier series are then given by

$$\begin{aligned}
a_0 &= \frac{1}{T} \int_0^T f(x) dx \\
&= \frac{1}{T} \int_0^{\frac{T}{2}} 1 dx + \frac{1}{T} \int_{\frac{T}{2}}^T -1 dx \\
&= 0 \\
a_n &= \frac{2}{T} \int_0^T f(x) \cos\left(\frac{2n\pi x}{T}\right) dx \\
&= \frac{2}{T} \int_0^{\frac{T}{2}} (1) \cos\left(\frac{2n\pi x}{T}\right) dx + \frac{2}{T} \int_{\frac{T}{2}}^T (-1) \cos\left(\frac{2n\pi x}{T}\right) dx \\
&= 0 \\
b_n &= \frac{2}{T} \int_0^T f(x) \sin\left(\frac{2n\pi x}{T}\right) dx
\end{aligned}$$

$$\begin{aligned}
&= \frac{2}{T} \int_0^{\frac{T}{2}} (1) \sin\left(\frac{2n\pi x}{T}\right) dx + \frac{2}{T} \int_{\frac{T}{2}}^T (-1) \sin\left(\frac{2n\pi x}{T}\right) dx \\
&= \begin{cases} \frac{2}{n\pi} & \text{if } n \text{ is even} \\ 0 & \text{if } n \text{ is odd} \end{cases}
\end{aligned}$$

The Fourier series is therefore given by

$$f(x) = \frac{4}{\pi} \sum_{n=1,3,5,\dots}^{\infty} \frac{1}{n} \sin\left(\frac{2n\pi x}{T}\right) \quad (3)$$

### (II.III) REGRESSION MODEL

(1) is a *linear model* for regression. Approximating the output values  $y_n$  using the linear model has an associated error. Minimizing this sum of squared errors results in an optimal set of weights. The error in approximation is given by  $(y_n - \hat{y}_n)^2$  where  $\hat{y}_n$  is the estimated  $y_n$  value. The combined error for all  $N$  data points can then be written as

$$\mathcal{E} = \sum_{n=1}^N (y_n - \hat{y}_n)^2 \quad (4)$$

The error given a  $M$  and  $\bar{w}$  is

$$\mathcal{E}(\bar{w}) = \sum_{n=1}^N (y_n - \sum_{i=1}^M w_i \phi_i(x_n))^2 \quad (5)$$

The optimal set of weights is one that minimizes  $\mathcal{E}(\bar{w})$

$$\bar{w}^* = \underset{\bar{w}}{\operatorname{argmin}} \mathcal{E}(\bar{w})$$

Differentiating  $\mathcal{E}(\bar{w})$  with respect to  $\bar{w}$ , we have

$$\begin{aligned}
\frac{d}{d\bar{w}} \mathcal{E}(\bar{w}) &= \frac{d}{d\bar{w}} \sum_{n=1}^N (y_n - \bar{w}^T \bar{\phi}(x_n))^2 \\
&= 2 \sum_{n=1}^N (y_n - \bar{\phi}(x_n)^T \bar{w}) \bar{\phi}(x_n)
\end{aligned}$$

Now,

$$\begin{aligned}
\frac{d}{d\bar{w}} \mathcal{E}(\bar{w}) = 0 &\Rightarrow \sum_{n=1}^N (y_n - \bar{\phi}(x_n)^T \bar{w}) \bar{\phi}(x_n) = 0 \\
\therefore \sum_{n=1}^N y_n \bar{\phi}(x_n) &= \sum_{n=1}^N \bar{\phi}(x_n)^T \bar{w} \bar{\phi}(x_n) \quad (6)
\end{aligned}$$

If  $\bar{y} = \begin{bmatrix} y_1 \\ y_2 \\ \vdots \\ y_N \end{bmatrix}$  and  $\Phi = \begin{bmatrix} \bar{\phi}(x_1)^T \\ \bar{\phi}(x_2)^T \\ \vdots \\ \bar{\phi}(x_N)^T \end{bmatrix}_{N \times M}$ , then (??) can be expressed as

$$\begin{aligned} \Phi \bar{w} &= \bar{y} \\ (\Phi^T \Phi) \bar{w} &= \Phi^T \bar{y} \end{aligned} \quad (7)$$

$$\bar{w} = (\Phi^T \Phi)^{-1} \Phi^T \bar{y} \quad (8)$$

The point of this derivation is to show that the optimal set of weights can be computed analytically. (7) is a system of linear equations. This value of  $\bar{w}$  corresponds to the regression fit for the given data samples. Given a basis set  $\bar{\phi}$  and the data values, one can construct the matrix  $\Phi$ . (7) can therefore be solved for  $\bar{w}$  by using Gaussian Elimination (using pivoting) or LU-Decomposition. Alternatively, one can solve for  $\bar{w}$  using (8) by computing the *pseudo inverse*  $(\Phi^T \Phi)^{-1}$ . Computing the inverse however is numerically unstable and hence, LU-Decomposition is usually preferred.

In this experiment,  $\bar{\phi} = [1 \sin x \cos x \sin 2x \cos 2x \dots \sin mx \cos mx]$  is an orthogonal basis set. Hence  $\Phi$  will be of the following form:

$$\Phi = \begin{bmatrix} 1 \sin x_1 \cos x_1 \sin 2x_1 \cos 2x_1 \dots \sin mx_1 \cos mx_1 \\ 1 \sin x_2 \cos x_2 \sin 2x_2 \cos 2x_2 \dots \sin mx_2 \cos mx_2 \\ \vdots \\ 1 \sin x_N \cos x_N \sin 2x_N \cos 2x_N \dots \sin mx_N \cos mx_N \end{bmatrix} \quad (9)$$

### (III) EXPERIMENT

The experiment involves coming up with the optimal number of terms in the orthogonal basis set to best approximate the underlying function. As the number of terms is varied, the hypothesis or the approximating function changes. As one increases the number of terms, the regression fit gets better i.e., the sum of squared errors decreases. The question that one needs to address is whether one can afford to increase the number of terms unconditionally.

**Proposition:-** The Minimum Message Length (MML) framework can provide a direction in answering the above question. I am using MML as a determining criterion to predict the optimal number of terms to be used in the approximating function. The whole point of this exercise is to determine the hypothesis (the number of terms plus their corresponding weights). This involves choosing the hypothesis which results in the least overall message length required to encode the hypothesis and the data given the hypothesis.

To verify the above proposition, data points are generated from a known function and noise is added to them. For a given number of terms, the

weights corresponding to the sum of least squares is computed. This would be the hypothesis that is used to approximate the data. The message length for this setting is then evaluated. This process is repeated for increasing value of number of terms and the value at which the length of the encoding message is minimum is regarded as the optimal number of terms. It is important to note that usually, the function which is used to generate the data is unknown..

For simulation purposes, the above tests are done by generating data from two functions, namely, the **sawtooth** and **square** functions shown in Figures 1 and 2 respectively. Each of these functions can be represented using an infinite Fourier series representation as detailed in (2) and (3) respectively.

### Data Generation

- *Generating X's*: For a given range and the number of samples, the  $x$  values are generated randomly as per a uniform distribution.
- For a chosen sawtooth/square function, the function values  $f(x)$  corresponding to the  $x$  values generated are then computed.
- *Generating Y's*: To the previously generated  $f(x)$  values, *Gaussian noise* is added to account for any errors in the actual experiment conducted.

$$y = f(x) + \epsilon \quad \text{and} \quad \epsilon \sim \mathcal{N}(\mu, \sigma)$$

where  $\mu$  and  $\sigma$  are the parameters of the Normal distribution and  $\epsilon$  is the amount of noise added.

### Parameters

The parameters for a model are specified during runtime. They are as follows:

- number of data samples ( $N$ )
- number of terms in the Fourier series under consideration ( $M$ )
- range (lower bound & upper bounds of the interval) from which the  $x$  values are generated
- the function according to which data is generated
- the mean ( $\mu$ ) and standard deviation ( $\sigma$ ) of the Gaussian noise
- time period of the chosen wave form ( $T$ )

### Regression fit

- For a given value of the number of terms the matrix  $\Phi$  is evaluated as shown in (9). This would be a  $N \times M$  rectangular matrix.
- The system of linear equations in (7) is constructed and solved for the weight vector  $\bar{w}$ .

- $\bar{w}$  corresponds to the linear regression fit for the model. Using this  $\bar{w}$ , the predictions ( $\hat{y}$ ) for the  $x$  values are computed using (1). The greater the number of terms, the better the fit in terms of minimizing the squared error (4).
- Figure 3 represents the regression fit using the first 3 terms  $\{1, \sin x, \cos x\}$  from the orthogonal basis set and Figure 4 represents the regression fit using the first 9 terms  $\{1, \sin x, \cos x, \dots, \sin 4x, \cos 4x\}$ . These two plots verify the fact that increasing the number of terms results in a better fit (minimized squared error).

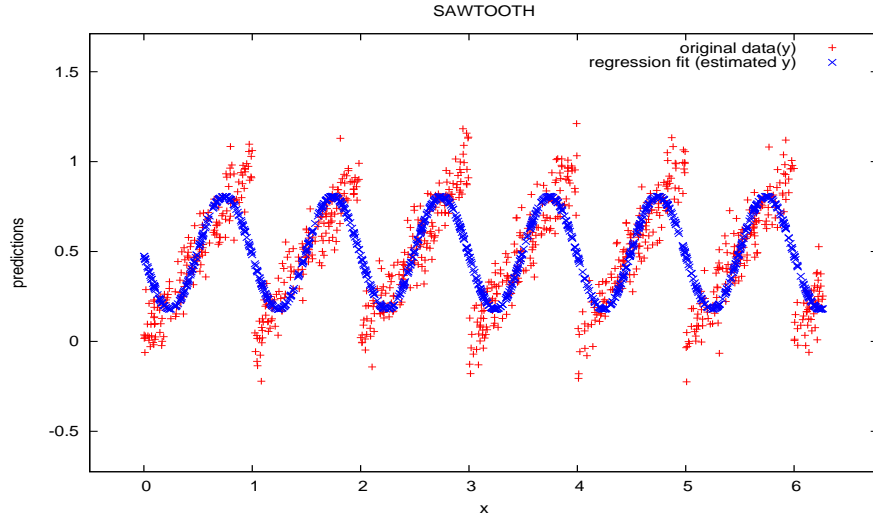


Figure 3: Regression fit for *sawtooth* wave using  $M = 3$  terms and  $\sigma = 0.1$

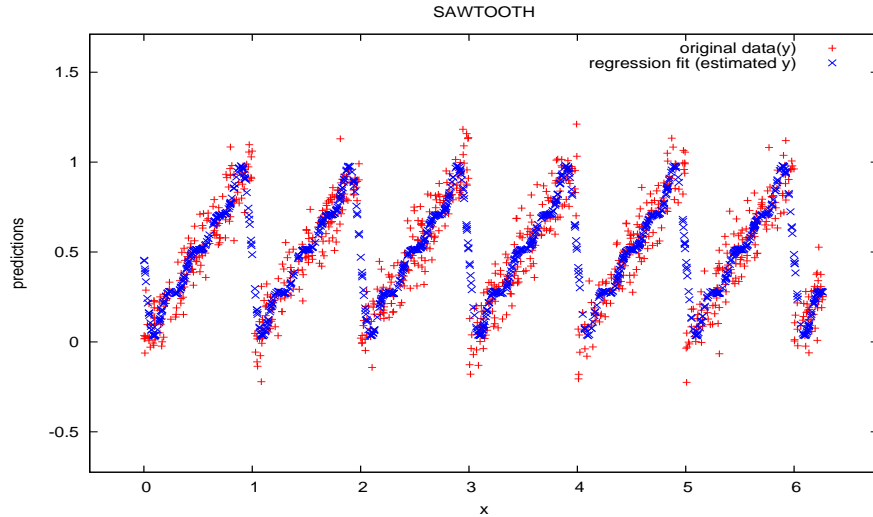


Figure 4: Regression fit for *sawtooth* wave using  $M = 9$  terms and  $\sigma = 0.1$

- Figure 5 shows the case when the data is generated without any noise. When the regression is done with 9 terms, for a sawtooth wave,

the weights (coefficients in the Fourier series representation (2)) are zero for the cosine terms and non-zero for the sine terms. Hence, the contribution is mainly due to  $1, \sin x, \sin 2x, \sin 3x, \sin 4x$  terms. That is the reason why we see 5 intermediate peaks (sub-waves – the blue curve) in the approximating function.

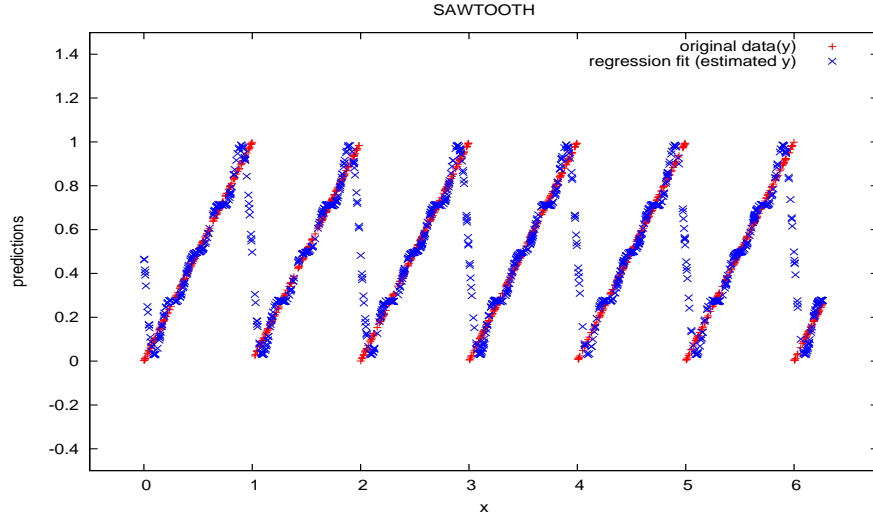


Figure 5: Regression fit for *sawtooth* wave using  $M = 9$  terms and  $\sigma = 0$

- Similar behaviour is observed for a square wave. Figure 6 represents the fit using 4 terms and Figure 7 represents a fit using 10 terms.

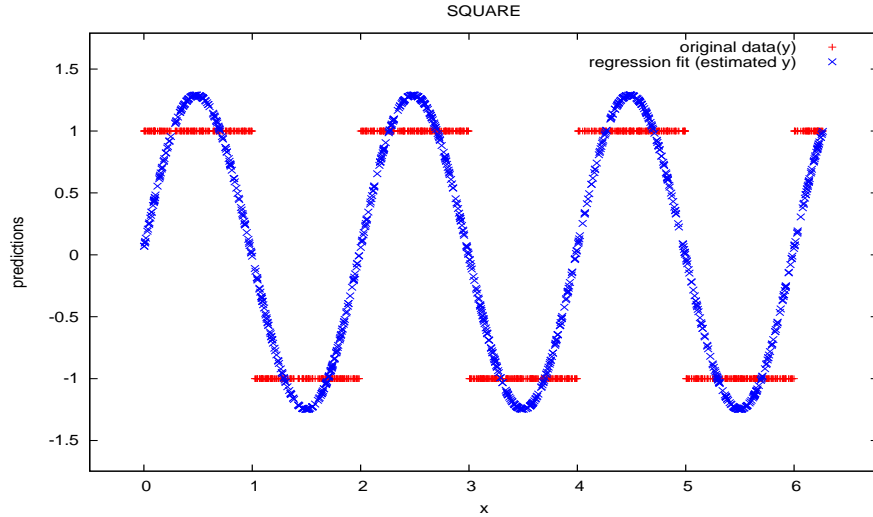


Figure 6: Regression fit for *square* wave using  $M = 4$  terms and  $\sigma = 0$

- For a square wave, the coefficients of its infinite Fourier series representation (shown in (3)) are non-zero for odd sines. Therefore, the main contribution is due to  $\sin x, \sin 3x, \sin 5x$  (3 peaks) terms when



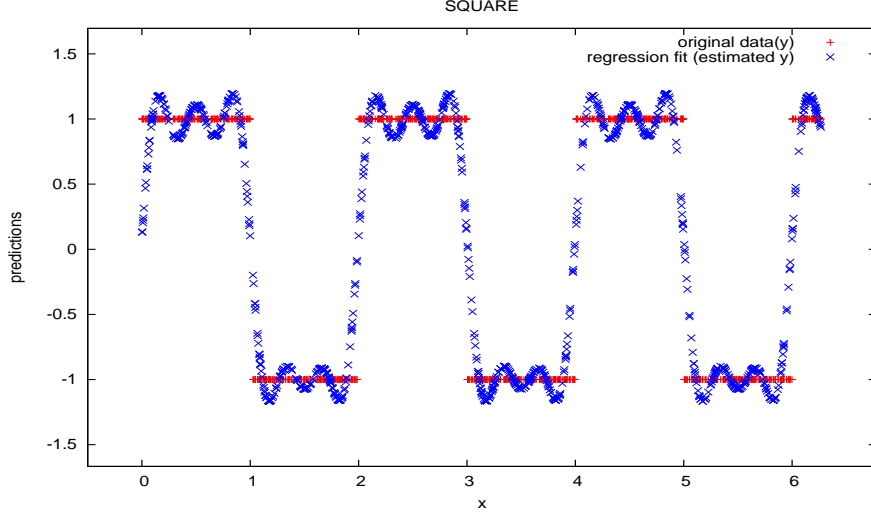


Figure 7: Regression fit for *square* wave using  $M = 10$  terms and  $\sigma = 0$

10 terms are used in the approximation. When 4 terms are used, the contribution is only due to the  $\sin x$  (just 1 peak) term. Hence in Figure 6, a distinct sine-like curve is observed.

### Computation of Message Length

- Message Length can be thought of as the length of the encryption a transmitter sends across. There is a receiver at the other end of the transmission channel and he decodes the encrypted message. Both the transmitter and the receiver adhere to a *codebook* which contains some basic things that both agree on. We are interested in lossless transmission and if the message is sent as per a Gaussian distribution, it is referred as a Gaussian channel.
- In a MML framework, for a given hypothesis, the message length is computed as the number of bits used in encoding the hypothesis and the data given the hypothesis. A message has two components. In this experiment, the hypothesis refers to the model in consideration. The model is defined by the number of terms and the weights. This is the first component. The second part of the message encodes the data given a particular model.
- To transmit a set of observations assuming that they are drawn from a Gaussian distribution, one needs to encode the Gaussian parameters ( $\mu$  and  $\sigma$ ), and then need to encode the observations using the distribution. This two part message when computed turns out to be the following expression using the Wallace Freeman approach.

$$\frac{1}{2} \log \frac{N\sigma^2}{N-1} + \frac{1}{2}(N-1) - \frac{1}{2}N \log \frac{2\pi}{\epsilon^2} + \frac{1}{2}(2N^2) + \log(R_\mu R_\sigma) + 1 + \log(K_2) \quad (10)$$

where  $N$  – number of observations  
 $\epsilon$  – accuracy of measurement  
 $R_\mu$ – range of mean of normal distribution  
 $R_\sigma$ – range of  $\log(\sigma)$  of normal distribution  
 $K_2$ – lattice constant

$R_\mu$  and  $R_\sigma$  are dependent on the type of problem one is dealing with. They are usually chosen based on the domain knowledge. As an example, if one is interested in encoding the heights of individuals, one can assume that average height to be between 5 and 6 feet.  $R_\mu$  would then be  $6-5 = 1$ . One could similarly come up with reasonable estimate for  $R_\sigma$ . In the current experiment, I have shown how to estimate these parameters in (11) and (15).

- *Encoding weights*: The transmitter needs to send the weights across to the receiver. It is assumed that the weights correspond to values sampled from a normal distribution and hence the message length used to encode these weights is computed as per (10). To calculate the message length, one needs to estimate the parameters of the distribution and also compute  $R_\mu$  and  $R_\sigma$  to be used in (10).

From the infinite Fourier series representation of sawtooth and square waveforms shown in (2) and (3), one can see that the coefficients (weights) of the sine and cosine terms are of the form  $\pm \frac{1}{n\pi}$ . This means the magnitude of the weights is always less than 1. This helps in determining the bounds for each weight  $w_i$ . Hence  $|w_i| < 1$  and  $R_\mu = 2$ .

To determine  $R_\sigma$ , one needs to estimate the bounds for  $\sigma$ . Now  $\sigma$  cannot be zero as it would mean the deviation from mean is zero and all observations are the same as the mean. Hence  $\sigma$  is bounded from below by  $\epsilon$ , the accuracy of measurement. The lower bound of  $\sigma$  is set to be a constant times  $\epsilon$ . The magnitude of the weights decreases as the number of terms increases in the Fourier series representation. The upper bound of  $\sigma$  is taken to be 1.

- *Encoding  $X$* : In the experiment, the  $x$  values are sampled from a pre-defined range  $[a, b]$ . Further, they are sorted in increasing order. The first term of this sorted sequence is made 0 and this part of the codebook. The other  $x$  values are scaled accordingly.

Instead of sending the  $x$ 's, what is sent is the difference  $\Delta x$  between consecutive  $x$  values. This will enable the receiver to construct the current  $x$  value using the previous  $x$  value received and the difference  $\Delta x$ . Sending  $\Delta x$ 's results in a compact message as well. Hence, information is sent in an efficient manner.  $\Delta x$ 's are sent over a Gaussian channel. The overall message length to encode  $\Delta x$ 's is computed as per (10). However, we still need to estimate  $R_\mu$  and  $R_\sigma$ .

For a set of  $x$ 's which are sampled from  $[a, b]$ , one can compute the sample mean and variance. Further, one can derive the bounds for the mean and standard deviation. The following is how I estimate  $R_\mu$  and  $R_\sigma$ .

1. To estimate  $R_{\mu_{\Delta x}}$

$$\begin{aligned} x \in [a, b] &\Rightarrow a \leq x \leq b \\ \therefore a \leq x_i \leq b \quad \text{and} \quad -b \leq x_j \leq -a \\ \text{If } \Delta x = x_i - x_j, \quad a - b \leq \Delta x \leq b - a \\ \therefore a - b \leq \mu_{\Delta x} \leq b - a \end{aligned} \quad (11)$$

The above result means if a set of observations are sampled from a known range, the mean of those observations also lies in that range.  $\Delta x \in [a - b, b - a]$ , and hence  $\mu_{\Delta x}$  also lies in that range. Hence  $R_{\mu_{\Delta x}} = 2(b - a)$

2. To estimate  $R_{\sigma_{\Delta x}}$

$$\sigma_{\Delta x}^2 = \sum_{i=1}^{N-1} \frac{(\Delta x_i - \mu_{\Delta x})^2}{N-1}$$

The following exercise is done to determine the bounds of  $\sigma_{\Delta x}$

$$\begin{aligned} \text{Consider } (\Delta x_i - \mu_{\Delta x})^2 &= \Delta x^2 + \mu_{\Delta x}^2 - 2\Delta x\mu_{\Delta x} \\ a - b \leq \mu_{\Delta x} \leq b - a &\Rightarrow 0 \leq \Delta x^2 \leq (b - a)^2 \end{aligned} \quad (12)$$

$$\text{and } a - b \leq \mu_{\Delta x} \leq b - a \Rightarrow 0 \leq \mu_{\Delta x}^2 \leq (b - a)^2 \quad (13)$$

$$\text{Also } -2(b - a)^2 \leq -2\Delta x\mu_{\Delta x} \leq 2(b - a)^2 \quad (14)$$

Adding (12), (13), (14) :-

$$\begin{aligned} -2(b - a)^2 &\leq (\Delta x_i - \mu_{\Delta x})^2 \leq 4(b - a)^2 \\ \therefore 0 &\leq \frac{(\Delta x_i - \mu_{\Delta x})^2}{N-1} \leq \frac{4(b - a)^2}{N-1} \\ 0 &\leq \sigma_{\Delta x}^2 \leq \frac{4(b - a)^2}{N-1} \\ \therefore 0 &\leq |\sigma_{\Delta x}| \leq \frac{2(b - a)}{\sqrt{N-1}} \end{aligned} \quad (15)$$

Equation (15) gives an upper bound on  $\log(\sigma_{\Delta x})$ . Since  $\sigma_{\Delta x}$  cannot be zero as it is a measure of the deviation from the mean, a lower bound is assumed for  $\sigma_{\Delta x}$  and is set to  $3\epsilon$  (a constant factor of the accuracy of measurement). Hence,  $R_{\sigma_{\Delta x}} = \log \frac{2(b-a)}{\sqrt{N-1}} - \log(3\epsilon)$ .

Using these values of  $R_{\mu_{\Delta x}}$  and  $R_{\sigma_{\Delta x}}$  in (10), the message length to encode  $\Delta x$  is computed.

- *Encoding Y*: The receiver can decode and infer the weights and  $x$  values sent by the transmitter. Further these two can be used to construct  $\hat{y}$  using (1). Since the transmitter knows that the receiver has access to this information, instead of sending the original  $y$  values, the difference  $\Delta y = \hat{y} - y$  is transmitted. The receiver can then construct the original  $y$  on his side of the channel. Transmitting the differences in  $y$  values is an efficient form of transmission as the encoding/decoding is done using previous knowledge and there is no redundancy in the data transmitted.

To compute the length of encoding the  $\Delta y$ 's, it is again assumed that they are derived from a Gaussian distribution and hence the message length follows (10).

#### (IV) OBSERVATIONS

- The above experiment is run for different values of number of data points ( $N$ ), and different values of Gaussian noise ( $\sigma$ ). Data is generated in the range  $[0, 2\pi]$  and time period of the wave forms set to 1. The below discussion is valid for both the sawtooth and the square waves.
- It should be expected that the message length continuously decreases as the number of terms ( $M$ ) increases, reaches a minimum, and then starts to increase. This is the ideal behaviour that one would expect.
- The theory is validated for  $N = 1000$  and  $N = 10000$  for varying values of  $\sigma$ . Please refer to the attached plots. The plots show the ideal behaviour. However the plots become erratic when the message lengths are plotted for  $N = 100$ .
- **An interesting observation:** The optimal value of  $M$  is the one which results in the overall minimum message length. It was interesting to note that the weights corresponding to a value of  $M$  are the corresponding coefficients in the Fourier expansion. But this is observed as long as the graph behaves ideally. There is a strong correlation between the erratic behaviour and the divergence of the inferred weights from the Fourier coefficients – I think this is a potential indicator of strange behaviour! (more about this is in the discussion below)

#### The $N = 100$ case:-

The following discussion is with respect to the sawtooth function.

- For a given value of  $\sigma$ , the graph tapers off at the beginning and then steadily increases until a certain value of  $M$ . From then on, the ideality is lost and the graph is erratic. It doesn't seem to follow a pattern from then on. It behaves randomly and at times, the message length increases and decreases with no clear indication as to why.

- Now I investigated as to why this was happening. I thought I will eliminate the most obvious choice first – problem with my code. There might be a problem with the way I am solving for  $\bar{w}$  in (7). I used three different techniques to do this.
  - I first tried the analytical solution, trying to compute  $\Phi^T \Phi^{-1}$  and solving for the weights analytically as in (8). I implemented the matrix inverse using Gaussian elimination (with partial pivoting).
  - Secondly, I used the `boost` library implementation of LU-Decomposition to solve for the matrix inverse.
  - Thirdly, I used my own implementation of LU-Decomposition to solve for the linear system (7).
  - **All the three methods produce the same results.** So there was nothing wrong with the way I was solving the linear system.
  - I did a few other sanity checks just to make sure that my code is not playing up. I computed the determinant using my implementation and compared it with `boost` implementation. Both of them agree.
  - I checked whether the  $\Phi$  matrix was being generated correctly. It seems to be the case.
  - I used this  $\Phi$  matrix and used `MATLAB` to compute  $\Phi^T \Phi$  and eventually computed the weights. These weights match the ones computed using my implementation. There however seem to be differences in  $\Phi^T \Phi^{-1}$  but they differ in the fractional part of the matrix elements. I think it is because of computational errors that creep in when dealing with large matrices. The weights computed are same nevertheless.
- For different values of  $M$ , I started printing out the corresponding weights and analysed them. Please refer to the plot for the sawtooth function ( $N = 100, \sigma = 0$ ). The graph shows ideal behaviour until about  $M = 55$  or so and then it starts behaving randomly. I printed the weights ranging from  $M = 1$  to  $M = 100$ . As long as the ideal behaviour was exhibited, the weights approximately match the Fourier coefficients. But they begin to diverge from then on. Consider the following specific values of  $M$ :
  - $M = 50$  – well behaved, everything is normal
  - at about  $M \sim 55$  – erratic behaviour begins
  - From  $M = 79$  to  $M = 80$ , the message length decreases. There is a reduction in the length of encoded message. This is bizarre as the message increases suddenly decreases and increases.

I've included the values of weights for these three cases in `comparison_msglen.txt` file for you to refer to. If you see, the weights for  $M = 50$  case match the Fourier coefficients. But when  $M = 79$  or  $M = 80$ , they are clearly far apart from their corresponding Fourier coefficients.

- I've also plotted the message length for part 1 (to encode weights) and part 2 (encoding data given weights). Please refer to the `comparison_msglen.eps` file. If you see there, the part 1 of the message is the one that is causing the problem. Part 1 of the overall message deals with encoding the weights. This forces me to think that there is something weird that is happening when the weights are being computed for higher value of  $M$ .
- Another important point to note is if you see the message length for part 2, it consistently keeps decreasing which points to the fact that the root mean squared error is steadily decreasing with increasing value of  $M$ . This is however expected because as the model complexity increases, the error of the regression fit decreases.
- Since this erratic behaviour is observed for  $N = 100$  data points, I was curious whether I can reproduce the similar behaviour with  $N = 1000$  data points. So I ran the experiment for 1000 data points (with  $\sigma = 0.1$ ) and let it run until the  $\Phi^T \Phi$  matrix becomes singular. As it turns out, this behaviour exists but for sufficiently large value of  $M$ . After about  $M = 400$  or so, the same bizzare behaviour appears and continues until  $M = 820$  when the matrix becomes singular and the program terminates. Please refer to this plot which I included in the folder.
- This forces me to think that the method breaks down for values of  $M$  which are comparable to the value of  $N$ .