

## Partial Differential Equation

Theodor Jonsson (thjo0148@student.umu.se)  
Simon Edvardsson (sied0020@student.umu.se)  
Mikkel Lindberg (jomi0065@student.umu.se)

## 1 Introduction

### 1.1 Computer lab signification

The computer lab as a whole revolves around numerical methods used to compute discrete Fourier transforms and extract the coefficients. The assignment consisted of 4 different tasks which test speed and accuracy.

### 1.2 Theory

All theory used to complete the four assignments are present in the computer lab description.

## 2 Tasks

### 2.1 Task 1

The first task asks to complete a Discrete Fourier Transform and Inverse Discrete Fourier Transform to transform a signal from time domain to frequency domain and back. Using the predetermined value of  $\mathbf{y} = (0, 3, -2\sqrt{3}, 6, 2\sqrt{3}, 3)$  and applying our implemented **mydft.m** to transform the signal to frequencies results in the following vector:

$$\mathbf{z} = (2, -\frac{1}{2} + i, \frac{1}{2} - i, -2, \frac{1}{2} + i, -\frac{1}{2} - i) = \mathbf{mydft}(\mathbf{y}).$$

Applying the inverse DFT on the same input vector  $\mathbf{y}$  yields:

$$\mathbf{w} = (12, -3 - 6i, 3 + 6i, -12, 3 - 6i, -3 + 6i) = \mathbf{myidft}(\mathbf{y}).$$

## 2.2 Task 2a

The goal of task 2a is to compute the Fourier coefficients for the given function:

$$f(x) = 3 - 2 * \cos(15x) + 4 * \sin(20x). \quad (1)$$

By inspection the resulting exact coefficients should be;  $a_0 = 3, a_{15} = -2, b_{20} = 4$ . When computing the Fourier coefficients using **myfouriercoeff.m** the following non zero coefficients are generated:

Table 1: Computed Fourier coefficients using  $N = 2^5$  data points.

$n$	$a_n$	$b_n$
0	3.0	0.0
12	0.0	-4.0
15	-2.0	0.0

Table 2: Computed Fourier coefficients using  $N = 2^8$  data points.

$n$	$a_n$	$b_n$
0	3.0	0.0
15	-2.0	0.0
20	0.0	4.0

As displayed in tables above the values of  $b_n$  is faulty when using  $N = 2^5$  individual data points in the interval. This is because there are not enough provided points to generate 20 Fourier sine coefficients. When using  $N$  data points a total of  $\frac{N}{2} - 1$  Fourier cosine coefficients along with  $\frac{N}{2} - 1$  Fourier sine coefficients and an additional constant term  $a_0$  will be generated. And since  $\frac{2^5}{2} - 1 = 15 \not\geq 20$  there will be no 20<sup>th</sup> cosine term.

## 2.3 Task 2b

Using the same function as in section 2.2 to compute the Fourier coefficients of the more intricate function:

$$g(x) = |\cos(x)| \quad x \in [0, 2\pi]. \quad (2)$$

Given that the Fourier coefficients can be expressed:

$$a_0 = \frac{2}{\pi}, \quad a_{2n} = -\frac{4}{\pi} \frac{(-1)^n}{4n^2 - 1}, \quad a_{2n+1} = 0, \quad b_n = 0 \quad (3)$$

Plotting the computed coefficients alongside the exact Fourier coefficients results in the following figure:

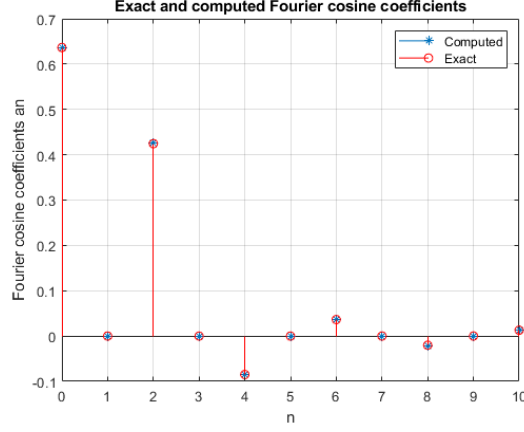


Figure 1: A plot of the computed Fourier coefficients and exact coefficients using equation (3) and  $2^6$  data points on the given interval.

## 2.4 Task 3a

The function given in the assignment is represented as:

$$h(x) = \begin{cases} 1 & 0 < x < \pi \\ -1 & \pi < x < 2\pi \\ 0 & x = 0 \text{ or } x = \pi \end{cases} \quad (4)$$

The function can be exactly represented by an infinite Fourier series but approximated with a finite number of Fourier coefficients. Using  $N = 2^8$  data points to approximate  $M$  coefficients along  $a_0$  results in the following plot:

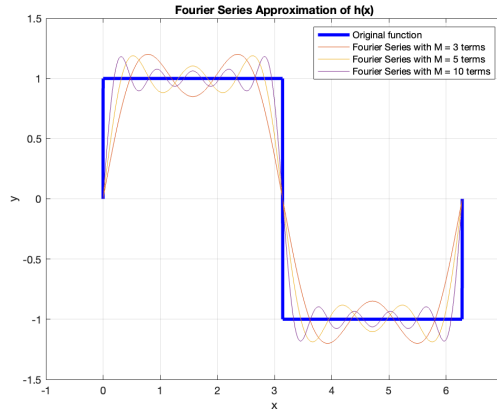


Figure 2: A plot of the computed Fourier series approximation of equation (4) using  $2^8$  data points on the given interval.

Using visual approximation, the overshoot of the Fourier series approximation above  $h(x) \equiv 1$  is slightly below 0.2 for  $M \in \{3, 5, 10\}$ . On the interval  $x \in (0, \pi)$ .

## 2.5 Task 3b

The task revolves around evaluating the overshoot of Fourier series approximation as  $M$  increases. Using  $M \in \{1, 2, \dots, 100\}$  coefficients to approximate  $h(x)$  and  $N = 2^{11}$  data points in the interval results in the following error,  $E(M)$ , development as  $M \rightarrow 100$ .

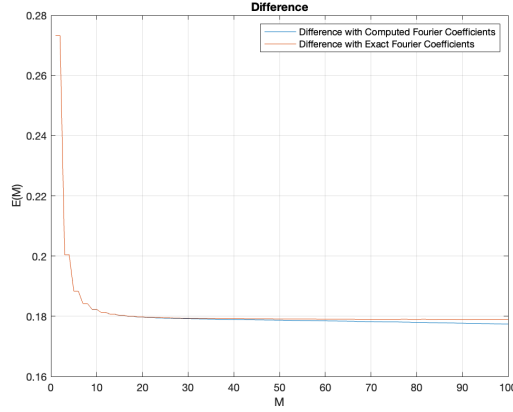


Figure 3: Figure of the difference between the Fourier series approximation.

Using the figure above it could be assumed that the difference between the Fourier series and the actual function  $h(x)$  using tapers of and approaches  $E(M) \approx 0.18$  as  $M \rightarrow \infty$ . Therefore there does not exist an  $M$  such that for  $\forall m \geq M$  then  $|f_m(x) - f(x)| < \epsilon$  for any  $\epsilon > 0$ . This means that the Fourier series does not satisfy the definition for uniform convergence.

## 2.6 Task 4a

The function **mydft** compared to the command FFT is slower and the time to run the function is significantly more affected by the size of the vector. The time for computing the function **mydft** gets exponential longer as the vector grows larger. While the value returned by the command FFT has a negligible change in time no matter the size of the vector.

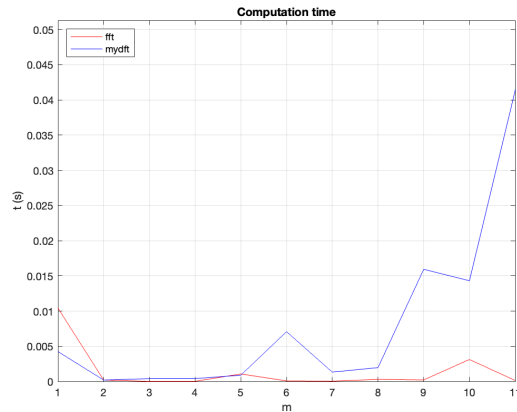


Figure 4: Figure of the computation time required for the Fast Fourier Transform (FFT) and the Discrete Fourier transform implemented in Task 1a under section ??.

The figure above displays the performance difference of the two methods, as FFT is significantly faster and seems to only require  $\mathcal{O}(n)$  time complexity relative to **mydft**.

## 2.7 Task 4b

When changing the threshold for the frequency, to ignore smaller frequencies, one can observe that the size of the compress file gets significant smaller. This is because the function compresses a bit of the data, and the output will then be smaller than the given input data.

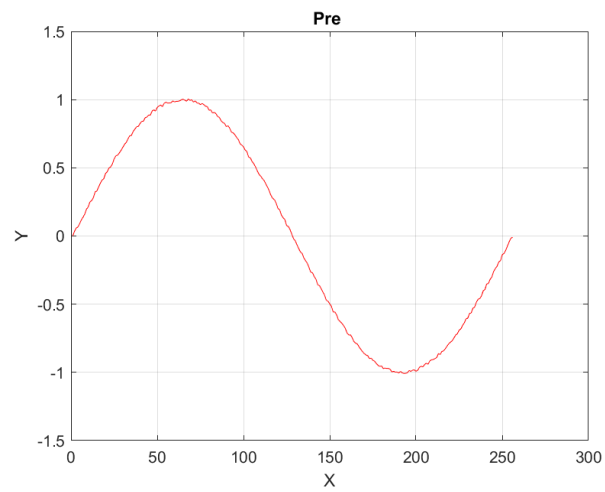


Figure 5: Figure of the unfiltered data.

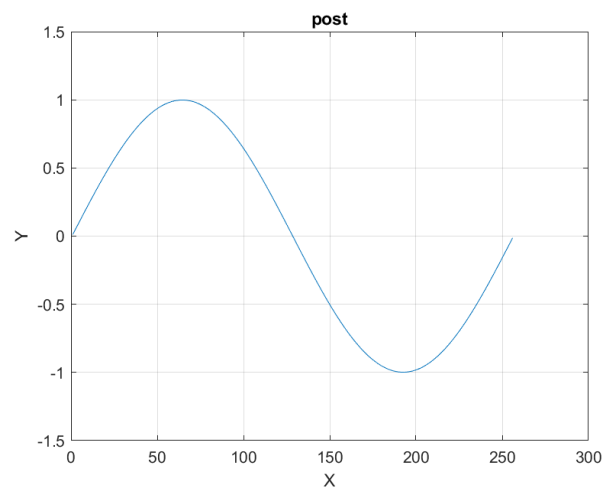


Figure 6: Figure of the filtered data.

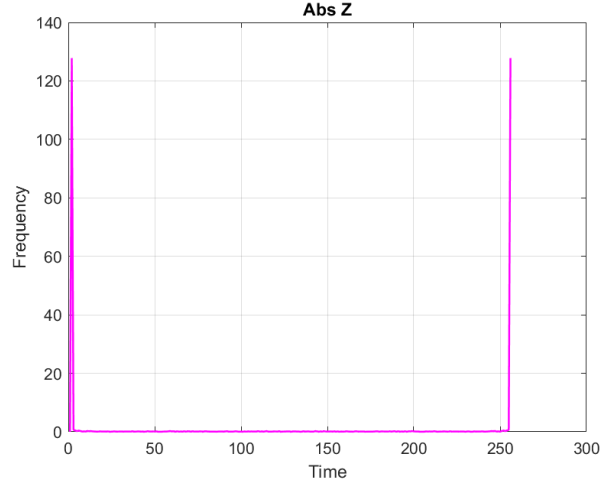


Figure 7: Figure of the absolute value of the Fourier transformed data of filtre.data. (file from lab description)

By comparing the two figures 5 and 6 the jaggedness of the unfiltered is removed and the figure comes out as a smooth curve as seen in figure 6. The value of  $w_{cut}$  which decides what frequencies should be cut or not is decided by plotting the absolute value of the Fourier transformed data from the filtre.data as seen in 7. A suitable  $w_{cut}$  is then chosen to be 10% of the maximum value of the transformed data to be sure to cut away all the low frequencies while keeping the high.

## 2.8 Task 4c

In task 4c a sound file is played chosen by a few different sounds in Matlab. The data of the sound is then transformed using FFT. The signal is then compressed using the following formula.

$$W(k) = \begin{cases} Y(k) & \text{if } |Y(k)| > \omega_r \cdot \max_k |Y(k)|, \\ 0 & \text{else.} \end{cases}$$

Where  $Y(k)$  is the Fourier transform of the given signal vector  $y$ , and  $\omega_r$  is a specified relative threshold of the maximum value of  $Y$ . The sound quality was then compared before and after the compression for a few different  $\omega_r$

Table 3: Train sound

$\omega_r$	compRatio	Quality
$\omega_{0.9}$	276 01	A single tone
$\omega_{0.8}$	193 21	Almost all sound gone.
$\omega_{0.5}$	772.84	Still recognisable but quite muffled
$\omega_{0.2}$	127.95	Almost the same as the original



Table 4: Splat sound

$\omega_r$	compRatio	Quality
$\omega_{0.9}$	750 12	No sound
$\omega_{0.8}$	272 77	No sound
$\omega_{0.5}$	43.549	Not recognisable but sounds like a cricket
$\omega_{0.2}$	5.1939	First half is the same but the other half quite muffled

Even though the  $\omega_r$  is the same, the compRatio differs from the train and splat sound. As can be seen from table (4) and (3) for example  $\omega_{0.8}$ , the sound splat has a much higher compRatio then for the train sound. But considering there is still sound coming from the train it must have frequencies with a higher amplitude.