Numerical Methods II Fourier Transforms and the FFT

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Outline

- Logistics
- 2 Trigonometric Orthogonal Polynomials
- Approximation Theory
- 4 Fast Fourier Transform
- Conclusions

Logistics

Course Essentials

- Course webpage: https://adonev.github.io/NumMethII
- Registered students: NYU Brightspace for announcements, submitting homeworks, grades, and sample solutions. Make sure you have access.
- Office hours: 3:00-4:30 pm Wednesdays, or by appointment, 1016 WWH.
 - TA's office hours (Mariya Savinov) office hours 2pm-3pm Mondays, 905 WWH.
- Main textbooks available in PDF format inside the NYU network/proxy: LeVeque, Trefethen (see course homepage).
- Other **optional readings** linked on course page; Ph.D. students should consult those sources.
- Computing is an essential part: **MATLAB** forms a common platform but (scientific/numerical) **Python/numpy** or **Julia** are great too.

To-do

- Assignment 0 is a Questionnaire with basic statistics about you: submit via email as plain text or PDF.
- Download PDFs of all textbooks for reference.
- There will be regular **homework assignments** (50% of grade), mostly computational. Points from all assignments will be added together.
- Submit the solutions as a PDF (give LaTex/lyx a try!), via NYU Brightspace. First assignment posted already and due in two weeks.
 - Due time is 5pm the day of class.
- You can choose between a take-home final or final project (50%), due TBD.

Academic Integrity Policy

- If you use any external source, even Wikipedia, make sure you acknowledge it by referencing all help.
- It is encouraged to discuss with other students the mathematical aspects, algorithmic strategy, code design, techniques for debugging, and compare results.
- Copying of any portion of someone else's solution or allowing others to copy your solution is considered cheating.
- Code sharing is not allowed: You must write/debug/run your own code.
- Submitting an individual and independent final is crucial and no collaboration will be allowed for the final.
- Common bad justifications for copying:
 - We are too busy and the homework is very hard, so we cannot do it on our own.
 - We do not copy each other but rather "work together."
 - I just emailed Joe Doe my solution as a "reference."

Grading Standards

- **Points** will be **added** over all assignments (50%) and the take-home final (50%).
- No makeup points (solutions may be posted on NYU Brightspace).
- The actual grades will be rounded upward (e.g., for those that are close to a boundary), but not downward:
 - 92.5-max = A
 - 87.5-92.5 = A-
 - 80.0-87.5 = B+
 - 72.5-80.0 = B
 - 65.0-72.5 = B-
 - 57.5-65.0 = C+
 - 50.0-57.5 = C
 - 42.5-50.0 = C-
 - min-42.5 = F

Trigonometric Orthogonal Polynomials

Periodic Functions

• Consider now interpolating / approximating **periodic functions** defined on the interval $I = [0, 2\pi]$:

$$\forall x \quad f(x+2\pi) = f(x),$$

as appear in practice when analyzing signals (e.g., sound/image processing).

• Also consider only the space of complex-valued square-integrable functions $L^2_{2\pi}$,

$$\forall f \in L_w^2: \quad (f,f) = ||f||^2 = \int_0^{2\pi} |f(x)|^2 dx < \infty.$$

- Polynomial functions are not periodic and thus basis sets based on orthogonal polynomials are not appropriate.
- Instead, consider sines and cosines as a basis function, combined together into complex exponential functions

$$\phi_k(x) = e^{ikx} = \cos(kx) + i\sin(kx), \quad k = 0, \pm 1, \pm 2, \dots$$

Fourier Basis Functions

$$\phi_k(x) = e^{ikx}, \quad k = 0, \pm 1, \pm 2, \dots$$

 It is easy to see that these are orhogonal with respect to the continuous dot product

$$(\phi_j, \phi_k) = \int_{x=0}^{2\pi} \phi_j(x) \phi_k^*(x) dx = \int_0^{2\pi} \exp[i(j-k)x] dx = 2\pi \delta_{ij}$$

• The complex exponentials can be shown to form a complete **trigonometric polynomial basis** for the space $L_{2\pi}^2$, i.e.,

$$orall f \in L^2_{2\pi}: \quad f(x) = \sum_{k=-\infty}^\infty \hat{f}_k e^{ikx}$$
 , in the sense of $L^2_{2\pi}$,

where the **Fourier coefficients** can be computed for any **frequency or wavenumber** k using:

$$\hat{f}_k = \frac{(f, \phi_k)}{2\pi} = \frac{1}{2\pi} \cdot \int_0^{2\pi} f(x) e^{-ikx} dx.$$

Truncated Fourier Basis

ullet For a general interval [0, X] the **discrete frequencies** are

$$k = \frac{2\pi}{X}\kappa$$
 $\kappa = 0, \pm 1, \pm 2, \dots$

- For non-periodic functions one can take the limit $X \to \infty$ in which case we get **continuous frequencies**.
- Now consider a discrete Fourier basis that only includes the first N
 basis functions, i.e.,

$$\begin{cases} k = -(N-1)/2, \dots, 0, \dots, (N-1)/2 & \text{if } N \text{ is odd} \\ k = -N/2, \dots, 0, \dots, N/2 - 1 & \text{if } N \text{ is even,} \end{cases}$$

and for simplicity we focus on N odd.

• The least-squares spectral approximation for this basis is:

$$f(x) \approx \phi(x) = \sum_{k=-(N-1)/2}^{(N-1)/2} \hat{f}_k e^{ikx}.$$

Discrete Dot Product

Now also discretize a given function on a set of N equi-spaced nodes

$$x_j = jh$$
 where $h = \frac{2\pi}{N}$

where j = N is the same node as j = 0 due to periodicity so we only consider N instead of N + 1 nodes.

• We also have the **discrete dot product** between two discrete functions (vectors) $\mathbf{f}_i = f(x_i)$:

$$\mathbf{f} \cdot \mathbf{g} = h \sum_{i=0}^{N-1} f_i g_i^*$$

• The discrete Fourier basis is discretely orthogonal

$$\phi_{\mathbf{k}} \cdot \phi_{\mathbf{k}'} = 2\pi \delta_{\mathbf{k},\mathbf{k}'}$$

Proof of Discrete Orthogonality

The case k = k' is trivial, so focus on

$$\phi_k \cdot \phi_{k'} = 0$$
 for $k \neq k'$

$$\sum_{j} \exp(ikx_{j}) \exp(-ik'x_{j}) = \sum_{j} \exp[i(\Delta k)x_{j}] = \sum_{j=0}^{N-1} [\exp(ih(\Delta k))]^{j}$$

where $\Delta k = k - k'$. This is a geometric series sum:

$$\phi_k \cdot \phi_{k'} = \frac{1 - z^N}{1 - z} = 0 \text{ if } k \neq k'$$

since
$$z = \exp(ih(\Delta k)) \neq 1$$
 and $z^N = \exp(ihN(\Delta k)) = \exp(2\pi i(\Delta k)) = 1$.

Discrete Fourier Transform

The Fourier interpolating polynomial is thus easy to construct

$$\phi^{(N)}(x) = \sum_{k=-(N-1)/2}^{(N-1)/2} \hat{f}_k^{(N)} e^{ikx}$$

where the discrete Fourier coefficients are given by

$$\hat{f}_k^{(N)} = \frac{\mathbf{f} \cdot \phi_k}{2\pi} = \frac{1}{N} \sum_{j=0}^{N-1} f(x_j) \exp(-ikx_j)$$

• Simplifying the notation and recalling $x_j = jh$, we define the the **Discrete Fourier Transform** (DFT):

$$\hat{f}_k^{(N)} = \frac{1}{N} \sum_{i=0}^{N-1} f_i \exp\left(-\frac{2\pi i j k}{N}\right)$$

Discrete spectrum

• The set of discrete Fourier coefficients $\hat{\mathbf{f}}$ is called the **discrete** spectrum, and in particular,

$$S_k = \left| \hat{f}_k \right|^2 = \hat{f}_k \hat{f}_k^{\star},$$

is the **power spectrum** which measures the frequency content of a signal.

• If f is real, then \hat{f} satisfies the **conjugacy property**

$$\hat{f}_{-k} = \hat{f}_k^{\star},$$

so that half of the spectrum is redundant and \hat{f}_0 is real.

• For an even number of points N the largest frequency k = -N/2 does not have a conjugate partner. It is special and **must be treated** with care.

Fourier Spectral Approximation

• Discrete Fourier Transform (DFT):

Forward
$$\mathbf{f} \rightarrow \hat{\mathbf{f}}: \quad \hat{f}_k = \frac{1}{N} \sum_{i=0}^{N-1} f_i \exp\left(-\frac{2\pi i j k}{N}\right)$$

Inverse
$$\hat{\mathbf{f}} o f$$
: $f(x_j) pprox \phi^{(N)}(x_j) = \sum_{k=-(N-1)/2}^{(N-1)/2} \hat{f}_k \exp\left(\frac{2\pi i j k}{N}\right)$

- There is a very fast algorithm for performing the forward and backward DFTs (FFT).
- There is **different conventions** for the DFT depending on the interval on which the function is defined and placement of factors of N and 2π .

Read the documentation to be consistent!

Spectral Convergence (or not)

• The Fourier interpolating polynomial $\phi(x)$ has **spectral accuracy**, i.e., exponential in the number of nodes N

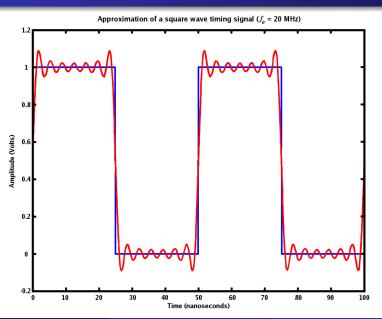
$$||f(x)-\phi^{(N)}(x)||\sim e^{-N}$$

for analytic functions (more details shortly).

- Specifically, nice functions exhibit rapid decay of the Fourier coefficients with k, e.g., exponential decay $\left|\hat{f}_k\right| \sim e^{-|k|}$.
- Discontinuities cause slowly-decaying Fourier coefficients, e.g., power law decay $\left|\hat{f}_k\right|\sim k^{-1}$ for **jump discontinuities**.
- Jump discontinuities lead to slow convergence of the Fourier series for non-singular points (and no convergence at all near the singularity), so-called Gibbs phenomenon (ringing):

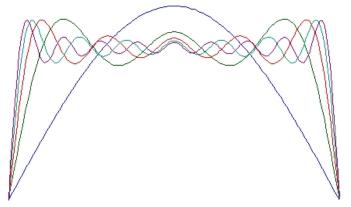
$$||f(x) - \phi^{(N)}(x)|| \sim \begin{cases} N^{-1} & \text{at points away from jumps} \\ \text{const.} & \text{at the jumps themselves} \end{cases}$$

Gibbs Phenomenon



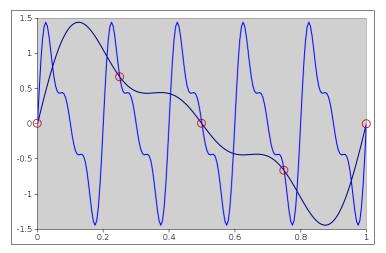
Gibbs Phenomenon

Reconstruction of the periodic square waveform with 1, 3, 5, 7, 9 sinusoids



Aliasing

If we sample a signal at too few points the Fourier interpolant may be wildly wrong: aliasing of frequencies k and 2k, 3k, ...



Approximation Theory

Trigonometric projection vs. interpolation

 I will temporarily switch to notation in paper on periodic chebfun in paper of Trefethen et al, assuming odd number of points for simplicity:

$$f\left(t\in[0,2\pi]
ight)$$
 discretized with $N=2n+1$ points $t_m=rac{2\pi m}{N}$

Trigonometric projection
$$(\phi(x))$$
: $f_n(t) = \sum_{k=-n}^n c_k e^{ikt}$ $(\phi(x))$

Trigonometric interpolant:
$$p_n(t) = \sum_{k=-n}^n \tilde{c}_k e^{ikt} \quad (\phi^{(N)}(x)).$$

• Aliasing means that one cannot distinguish two different Fourier modes on a given grid:

$$\exp(ikt_m) = \exp(i(k+jN)t_m)$$

Poisson Summation Formula

Observe that because of aliasing:

$$\begin{split} f\left(t_{m}\right) &= \sum_{k=-\infty}^{\infty} c_{k} e^{ikt_{m}} = \sum_{k=-n}^{n} \sum_{j=-\infty}^{\infty} c_{k+jN} e^{i(k+jN)t_{m}} \\ &= \sum_{k=-n}^{n} \left(\sum_{j=-\infty}^{\infty} c_{k+jN}\right) e^{ikt_{m}} \\ \text{Recall: } p_{n}\left(t_{m}\right) &= \sum_{k=-n}^{n} \tilde{c}_{k} e^{ikt_{m}} \end{split}$$

 Since the trigonometric interpolant is unique, we get Poisson's summation formula

$$\tilde{c}_k = \sum_{j=-\infty}^{\infty} c_{k+jN}$$

The importance of smoothness

Total variation of differentiable function (can be generalized):

$$\mathsf{TV}[f] = \int_0^{2\pi} \left| f'(x) \right| dx, \quad \mathsf{denote} \ V = \mathsf{TV}\left[f^{(\nu)} \right].$$

- We have two cases where we have nice error estimates:
 - If f is $\nu \geq 0$ times **differentiable**, then

$$|c_k| \leq \frac{V}{2\pi \left|k\right|^{\nu+1}}$$

which can be proved by integrating $c_k = (2\pi)^{-1} \int_0^{2\pi} f(x) e^{-ikx} dx$ by parts $\nu + 1$ times.

• If f(t) is **analytic** in a half-strip around the real axis of half-width α and bounded by |f(t)| < M, then

$$|c_k| \leq Me^{-\alpha|k|}$$

which can be proved by shifting the contour of integration above or below the real line by α .

Approximation error: Differentiable

• If f is $\nu > 1$ times **differentiable** then

$$||f - f_n||_{\infty} = \left\| \sum_{|k| > n} c_k e^{ikt} \right\|_{\infty} \le \sum_{|k| > n} |c_k|$$

$$\le 2 \sum_{k=n+1}^{\infty} \frac{V}{2\pi k^{\nu+1}} \approx 2 \int_{n}^{\infty} \frac{V}{2\pi k^{\nu+1}} dk$$

• Performing the integral we get that if f is $\nu \geq 1$ times **differentiable**, then

$$||f - f_n||_{\infty} \le \frac{V}{\pi \nu n^{\nu}}$$

• You can replace f_n with p_n if you multiply the r.h.s. by 2 to account for the additional aliasing error.

Approximation error: Analytic

• If f(t) is **analytic** in the half strip then

$$\|f - f_n\|_{\infty} \le 2 \sum_{k=n+1}^{\infty} M e^{-\alpha k} = \frac{2M e^{-\alpha n}}{e^{\alpha} - 1}$$
 (geometric series sum)

- You can replace f_n with p_n if you multiply the r.h.s. by 2 to account for the additional aliasing error.
- The Fourier interpolating trigonometric polynomial is spectrally accurate and a really great approximation for (very) smooth functions.

Trapezoidal Rule

 Consider using the trapezoidal rule to approximate a periodic integral:

$$I=\int_0^{2\pi}f(x)dx=c_0$$

$$I_N = \frac{2\pi}{N} \sum_{m=1}^N f(t_m) = \tilde{c}_0.$$

• If f is $\nu \geq 1$ times **differentiable** then

$$|I_N - I| \leq \frac{4V}{N^{\nu+1}}.$$

 If f (t) is analytic in the half strip then trapezoidal rule is spectrally accurate:

$$|I_N - I| \le \frac{4\pi M}{e^{\alpha N} - 1}.$$

Fast Fourier Transform

DFT

 Recall the transformation from real space to frequency space and back:

$$\mathbf{f} \to \hat{\mathbf{f}} : \quad \hat{f}_k = \frac{1}{N} \sum_{j=0}^{N-1} f_j \exp\left(-\frac{2\pi i j k}{N}\right), \quad k = -\frac{(N-1)}{2}, \dots, \frac{(N-1)}{2}$$

$$\hat{\mathbf{f}}
ightarrow \mathbf{f}: \quad f_j = \sum_{k=-(N-1)/2}^{(N-1)/2} \hat{f}_k \exp\left(rac{2\pi i j k}{N}
ight), \quad j=0,\dots,N-1$$

• We can make the forward-reverse **Discrete Fourier Transform** (DFT) more symmetric if we shift the frequencies to k = 0, ..., N:

Forward
$$\mathbf{f} \to \hat{\mathbf{f}}$$
: $\hat{f}_k = \frac{1}{\sqrt{N}} \sum_{j=0}^{N-1} f_j \exp\left(-\frac{2\pi i j k}{N}\right), \quad k = 0, \dots, N-1$

Inverse
$$\hat{\mathbf{f}} \to \mathbf{f}$$
: $f_j = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} \hat{f}_k \exp\left(\frac{2\pi i j k}{N}\right), \quad j = 0, \dots, N-1$

FFT

• We can write the transforms in matrix notation:

$$\hat{\mathbf{f}} = \frac{1}{\sqrt{N}} \mathbf{U}_N \mathbf{f}$$
 $\mathbf{f} = \frac{1}{\sqrt{N}} \mathbf{U}_N^* \hat{\mathbf{f}},$

where the **unitary Fourier matrix** (fft(eye(N))) in MATLAB) is an $N \times N$ matrix with entries

$$u_{ik}^{(N)} = \omega_N^{jk}, \quad \omega_N = e^{-2\pi i/N}.$$

- A **direct** matrix-vector multiplication algorithm therefore takes $O(N^2)$ multiplications and additions.
- Is there a faster way to compute the non-normalized

$$\hat{f}_k = \sum_{j=0}^{N-1} f_j \omega_N^{jk} \quad ?$$

FFT.

- For now assume that N is even and in fact a power of two, $N = 2^n$.
- The idea is to split the transform into two pieces, even and odd points:

$$\sum_{j=2j'} f_j \omega_N^{jk} + \sum_{j=2j'+1} f_j \omega_N^{jk} = \sum_{j'=0}^{N/2-1} f_{2j'} \left(\omega_N^2\right)^{j'k} + \omega_N^k \sum_{j'=0}^{N/2-1} f_{2j'+1} \left(\omega_N^2\right)^{j'k}$$

Now notice that

$$\omega_N^2 = e^{-4\pi i/N} = e^{-2\pi i/(N/2)} = \omega_{N/2}$$

• This leads to a divide-and-conquer algorithm:

$$\begin{split} \hat{f}_k &= \sum_{j'=0}^{N/2-1} f_{2j'} \omega_{N/2}^{j'k} + \omega_N^k \sum_{j'=0}^{N/2-1} f_{2j'+1} \omega_{N/2}^{j'k} \\ \hat{f}_{k \geq 0} &= \mathbf{U}_{N/2} \mathbf{f}_{even} + \omega_N^{k \geq 0} \mathbf{U}_{N/2} \mathbf{f}_{odd} \\ \hat{f}_{k < 0} &= \mathbf{U}_{N/2} \mathbf{f}_{even} + \omega_N^{k < 0} \mathbf{U}_{N/2} \mathbf{f}_{odd} \end{split}$$

FFT Complexity

• The Fast Fourier Transform algorithm is recursive (in standard ordering $\mathbf{k} = [-(N-1)/2 : (N-1)/2]$:

$$\begin{aligned} \textit{FFT}_{\textit{N}}(\mathbf{f}) &= \left[\textit{FFT}_{\frac{N}{2}}(\mathbf{f}_{even}) + \mathbf{w}_{k<0} \boxdot \textit{FFT}_{\frac{N}{2}}(\mathbf{f}_{odd}), \\ \textit{FFT}_{\frac{N}{2}}(\mathbf{f}_{even}) + \mathbf{w}_{k\geq0} \boxdot \textit{FFT}_{\frac{N}{2}}(\mathbf{f}_{odd})\right] \end{aligned}$$

where $w_k = \omega_N^k$ and \boxdot denotes element-wise product. When N = 1 the FFT is trivial (identity).

- To compute the whole transform we need $log_2(N)$ steps, and at each step we only need N multiplications and N additions at each step.
- The total **cost of FFT** is thus much better than the direct method's $O(N^2)$: **Log-linear**

$$O(N \log N)$$
.

• Even when *N* is not a power of two there are ways to do a similar **splitting** transformation of the large FFT into many smaller FFTs.

In MATLAB

- The forward transform is performed by the function $\hat{f} = fft(f)$ and the inverse by $f = fft(\hat{f})$. Note that ifft(fft(f)) = f and f and \hat{f} may be complex. MATLAB uses the FFTW library, as does numpy.
- In MATLAB, and other software, the frequencies are not ordered in the "normal" way -(N-1)/2 to +(N-1)/2, but rather, the nonnegative frequencies come first, then the positive ones, so the "funny" ordering is

$$0,1,\ldots,(N-1)/2, \quad -\frac{N-1}{2},-\frac{N-1}{2}+1,\ldots,-1.$$

This is because such ordering (shift) makes the forward and inverse transforms symmetric, and reduces the amount of memory traffic.

• The function *fftshift* can be used to order the frequencies in the "normal" way, and *ifftshift* does the reverse:

$$\hat{f} = fftshift(fft(f))$$
 (normal ordering).

 Note that there are different normalization conventions used in different software.

FFT-based noise filtering (1)

```
Fs = 1000:
                               % Sampling frequency
dt = 1/Fs;
                               % Sampling interval
L = 1000:
                               % Length of signal
t = (0:L-1)*dt;
                               % Time vector
T=L*dt:
                               % Total time interval
% Sum of a 50 Hz sinusoid and a 120 Hz sinusoid
x = 0.7*sin(2*pi*50*t) + sin(2*pi*120*t);
y = x + 2*randn(size(t)); % Sinusoids plus noise
figure (1); clf;
plot(t(1:100), y(1:100), 'b--'); hold on
title ('Signal Corrupted with Zero-Mean Random Noise')
xlabel('time')
```

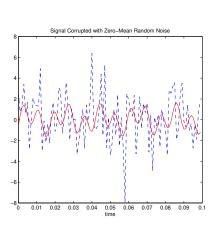
FFT-based noise filtering (2)

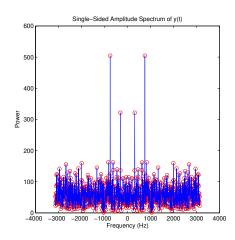
```
if (0)
   N=(L/2)*2; % Even N
   y_hat = fft(y(1:N));
  % Frequencies ordered in a funny way:
   f_{\text{funny}} = 2 * pi/T * [0:N/2-1, -N/2:-1];
  % Normal ordering:
   f_{normal} = 2*pi/T*[-N/2 : N/2-1];
else
   N=(L/2)*2-1; % Odd N
   y_hat = fft(y(1:N));
  % Frequencies ordered in a funny way:
   f_-funny = 2*pi/T* [0:(N-1)/2, -(N-1)/2:-1];
  % Normal ordering:
   f_{\text{normal}} = 2 * pi/T * [-(N-1)/2 : (N-1)/2];
end
```

FFT-based noise filtering (3)

```
figure (2); clf; plot(f_funny, abs(y_hat), 'ro');
y_hat=fftshift(y_hat);
figure (2); plot (f_normal, abs(y_hat), 'b-');
title ('Single-Sided Amplitude Spectrum of y(t)')
xlabel('Frequency (Hz)')
ylabel ('Power')
y_hat(abs(y_hat) < 250) = 0; \% Filter out noise
y_filtered = ifft(ifftshift(y_hat));
figure (1); plot (t(1:100), y_filtered(1:100), 'r-')
```

FFT results





Multidimensional FFT

 DFTs and FFTs generalize straightforwardly to higher dimensions due to separability: Transform each dimension independently

$$\hat{f} = \frac{1}{N_x N_y} \sum_{j_v=0}^{N_y-1} \sum_{j_x=0}^{N_x-1} f_{j_x,j_y} \exp\left[-\frac{2\pi i \left(j_x k_x + j_y k_y\right)}{N}\right]$$

$$\hat{\mathbf{f}}_{k_x, k_y} = \frac{1}{N_x} \sum_{j_y=0}^{N_y-1} \exp\left(-\frac{2\pi i j_y k_x}{N}\right) \left[\frac{1}{N_y} \sum_{j_y=0}^{N_y-1} f_{j_x, j_y} \exp\left(-\frac{2\pi i j_y k_y}{N}\right) \right]$$

• For example, in two dimensions, do FFTs of each column, then FFTs of each row of the result:

$$\hat{\mathbf{f}} = \boldsymbol{\mathcal{F}}_{row}\left(\boldsymbol{\mathcal{F}}_{col}\left(\mathbf{f}
ight)
ight)$$

• The cost is N_y one-dimensional FFTs of length N_x and then N_x one-dimensional FFTs of length N_y :

$$N_x N_y \log N_x + N_x N_y \log N_y = N_x N_y \log (N_x N_y) = N \log N$$

Conclusions

Conclusions/Summary

- Periodic functions can be approximated using basis of orthogonal trigonometric polynomials.
- The Fourier basis is discretely orthogonal and gives spectral accuracy for smooth functions.
- Functions with discontinuities are not approximated well: Gibbs phenomenon.
- The **Discrete Fourier Transform** can be computed very efficiently using the **Fast Fourier Transform** algorithm: $O(N \log N)$.