Digital sound and discrete Fourier analysis

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For complex vectors of length ${\it N}$ the Euclidean inner product is given by

$$\langle \boldsymbol{x}, \boldsymbol{y}
angle = \sum_{k=0}^{N-1} x_k \overline{y_k}.$$

The associated norm is

$$\|\mathbf{x}\| = \sqrt{\sum_{k=0}^{N-1} |x_k|^2}.$$

In Discrete Fourier analysis, a vector $\mathbf{x} = (x_0, \dots, x_{N-1})$ is represented as a linear combination of the *N* vectors

$$\phi_n = rac{1}{\sqrt{N}} \left(1, e^{2\pi i n/N}, e^{2\pi i 2n/N}, \dots, e^{2\pi i k n/N}, \dots, e^{2\pi i n(N-1)/N}
ight).$$

These vectors are called the normalised complex exponentials, or the pure digital tones of order *N*. *n* is also called frequency index. The whole collection $\mathcal{F}_N = \{\phi_n\}_{n=0}^{N-1}$ is called the *N*-point Fourier basis.

the *N*-point Fourier basis is an orthonormal basis for \mathbb{R}^N .

We will denote the change of coordinates matrix from the standard basis of \mathbb{R}^N to the Fourier basis \mathcal{F}_N by \mathcal{F}_N . We will also call this the (*N*-point) Fourier matrix.

The matrix $\sqrt{N}F_N$ is also called the (*N*-point) *discrete Fourier* transform, or DFT. If \mathbf{x} is a vector in \mathbb{R}^N , then $\mathbf{y} = \text{DFT}\mathbf{x}$ are called the DFT coefficients of \mathbf{x} . (the DFT coefficients are thus the coordinates in \mathcal{F}_N , scaled with \sqrt{N}). DFT \mathbf{x} is sometimes written as $\hat{\mathbf{x}}$.

Theorem 2.5: The Fourier matrix F_N is the unitary $N \times N$ -matrix with entries given by

$$(F_N)_{nk} = rac{1}{\sqrt{N}}e^{-2\pi i nk/N},$$

for $0 \leq n, k \leq N - 1$.

Definition 2.6: The matrix $\overline{F_N}/\sqrt{N}$ is the inverse of the matrix DFT = $\sqrt{N}F_N$. We call this inverse matrix the *inverse discrete* Fourier transform, or IDFT.

```
function y = DFTImpl(x)
N = size(x, 1);
y = zeros(size(x));
for n = 1:N
D = exp(2*pi*1i*(n-1)*(0:(N-1))/N);
y(n) = dot(D, x);
end
```

n has been replaced by n - 1 in this code since *n* runs from 1 to *N* (array indices must start at 1 in Matlab).

Direct implementation of the DFT in Python

```
def DFTImpl(x):
    y = zeros_like(x).astype(complex)
    N = len(x)
    for n in xrange(N):
        D = exp(-2*pi*n*1j*arange(float(N))/N)
        y[n] = dot(D, x)
    return y
```

Let \boldsymbol{x} be a real vector of length N. The DFT has the following properties:

$$(\widehat{\boldsymbol{x}})_{N-n} = \overline{(\widehat{\boldsymbol{x}})_n} \text{ for } 0 \le n \le N-1.$$

- **2** If $x_k = x_{N-k}$ for all n (so \mathbf{x} is symmetric), then $\hat{\mathbf{x}}$ is a real vector.
- **3** If $x_k = -x_{N-k}$ for all k (so \mathbf{x} is antisymmetric), then $\hat{\mathbf{x}}$ is a purely imaginary vector.
- If *d* is an integer and *z* is the vector with components $z_k = x_{k-d}$ (the vector *x* with its elements delayed by *d*), then $(\hat{z})_n = e^{-2\pi i dn/N} (\hat{x})_n$.
- If *d* is an integer and *z* is the vector with components $z_k = e^{2\pi i dk/N} x_k$, then $(\hat{z})_n = (\hat{x})_{n-d}$.

Relation between Fourier coefficients and DFT coefficients, Proposition 2.9

Let N > 2M, $f \in V_{M,T}$, and let $\mathbf{x} = \{f(kT/N)\}_{k=0}^{N-1}$ be N uniform samples from f over [0, T]. The Fourier coefficients z_n of f can be computed from

$$(z_0, z_1, \ldots, z_M, \underbrace{0, \ldots, 0}_{N-(2M+1)}, z_{-M}, z_{-M+1}, \ldots, z_{-1}) = \frac{1}{N} \mathsf{DFT}_N \mathbf{x}.$$

In particular, the total contribution in f from frequency n/T, for $0 \le n \le M$, is given by y_n and y_{N-n} , where y is the DFT of x.

Proposition 2.12: Any $f \in V_{M,T}$ can be reconstructed uniquely from a uniform set of samples $\{f(kT/N)\}_{k=0}^{N-1}$, as long as $f_s > 2|\nu|$, where ν denotes the highest frequency in f.

Let f be a periodic function with period T, and assume that f has no frequencies higher than ν Hz. Then f can be reconstructed exactly from its samples $f(-MT_s), \ldots, f(MT_s)$ (where T_s is the sampling period, $N = \frac{T}{T_s}$ is the number of samples per period, and M = 2N + 1) when the sampling rate $f_s = \frac{1}{T_s}$ is bigger than 2ν . Moreover, the reconstruction can be performed through the formula

$$f(t) = \sum_{k=-M}^{M} f(kT_s) \frac{1}{N} \frac{\sin(\pi(t-kT_s)/T_s)}{\sin(\pi(t-kT_s)/T)}.$$

Assume that f has no frequencies higher than ν Hz. Then f can be reconstructed exactly from its samples $\dots, f(-2T_s), f(-T_s), f(0), f(T_s), f(2T_s), \dots$ when the sampling rate is bigger than 2ν . Moreover, the reconstruction can be performed through the formula

$$f(t) = \sum_{k=-\infty}^{\infty} f(kT_s) \frac{\sin(\pi(t-kT_s)/T_s)}{\pi(t-kT_s)/T_s}.$$

Using the DFT to adjust frequencies in sound, Example 2.16

```
[x, fs] = forw_comp_rev_DFT('L', 13000, 'lower', 1);
playerobj=audioplayer(x, fs);
playblocking(playerobj);
```

```
[x, fs] = forw_comp_rev_DFT('threshold', 20);
playerobj=audioplayer(x, fs);
playblocking(playerobj);
```

Compression by quantizing DFT coefficients, Example 2.18

[x, fs] = forw_comp_rev_DFT('n', 3); playerobj=audioplayer(x, fs); playblocking(playerobj);

FFT algorithm when N is even Theorem 2.15

Let $\mathbf{y} = \text{DFT}_N \mathbf{x}$ be the *N*-point DFT of \mathbf{x} , with *N* an even number, and let $D_{N/2}$ be the $(N/2) \times (N/2)$ -diagonal matrix with entries $(D_{N/2})_{n,n} = e^{-2\pi i n/N}$ for $0 \le n < N/2$. Then we have that

$$(y_0, y_1, \dots, y_{N/2-1}) = \mathsf{DFT}_{N/2} \mathbf{x}^{(e)} + D_{N/2} \mathsf{DFT}_{N/2} \mathbf{x}^{(o)}$$
$$(y_{N/2}, y_{N/2+1}, \dots, y_{N-1}) = \mathsf{DFT}_{N/2} \mathbf{x}^{(e)} - D_{N/2} \mathsf{DFT}_{N/2} \mathbf{x}^{(o)}$$

where $\mathbf{x}^{(e)}, \mathbf{x}^{(o)} \in \mathbb{R}^{N/2}$ consist of the even- and odd-indexed entries of \mathbf{x} , respectively, i.e.

$$\mathbf{x}^{(e)} = (x_0, x_2, \dots, x_{N-2})$$
 $\mathbf{x}^{(o)} = (x_1, x_3, \dots, x_{N-1}).$

IFFT algorithm when N is even, Theorem 2.15

Let N be an even number and let $\tilde{x} = \overline{\text{DFT}_N y}$. Then we have that

$$\begin{aligned} & (\tilde{x}_0, \tilde{x}_1, \dots, \tilde{x}_{N/2-1}) = \overline{\mathsf{DFT}_{N/2}} \boldsymbol{y}^{(e)} + \overline{D_{N/2}\mathsf{DFT}_{N/2}}) \boldsymbol{y}^{(o)} \\ & (\tilde{x}_{N/2}, \tilde{x}_{N/2+1}, \dots, \tilde{x}_{N-1}) = \overline{\mathsf{DFT}_{N/2}} \boldsymbol{y}^{(e)} - \overline{D_{N/2}\mathsf{DFT}_{N/2}}) \boldsymbol{y}^{(o)} \end{aligned}$$

where $\pmb{y}^{(e)}, \pmb{y}^{(o)} \in \mathbb{R}^{N/2}$ are the vectors

$$\mathbf{y}^{(e)} = (y_0, y_2, \dots, y_{N-2})$$
 $\mathbf{y}^{(o)} = (y_1, y_3, \dots, y_{N-1}).$

Moreover, $\mathbf{x} = IDFT_N \mathbf{y}$ can be computed from $\mathbf{x} = \tilde{\mathbf{x}}/N = \overline{DFT_N} \mathbf{y}/N$

We have that

$$DFT_{N}\boldsymbol{x} = \begin{pmatrix} I & D_{N/2} \\ I & -D_{N/2} \end{pmatrix} \begin{pmatrix} DFT_{N/2} & \boldsymbol{0} \\ \boldsymbol{0} & DFT_{N/2} \end{pmatrix} \begin{pmatrix} \boldsymbol{x}^{(e)} \\ \boldsymbol{x}^{(o)} \end{pmatrix}$$
$$IDFT_{N}\boldsymbol{y} = \frac{1}{N} \overline{\begin{pmatrix} I & D_{N/2} \\ I & -D_{N/2} \end{pmatrix}} \begin{pmatrix} \overline{DFT_{N/2}} & \boldsymbol{0} \\ \boldsymbol{0} & \overline{DFT_{N/2}} \end{pmatrix} \begin{pmatrix} \boldsymbol{y}^{(e)} \\ \boldsymbol{y}^{(o)} \end{pmatrix}$$

Iterating the factorization 1

$$DFT_{N}\boldsymbol{x} = \begin{pmatrix} I & D_{N/2} \\ I & -D_{N/2} \end{pmatrix} \begin{pmatrix} I & D_{N/4} & \mathbf{0} & \mathbf{0} \\ I & -D_{N/4} & \mathbf{0} & \mathbf{0} \\ \mathbf{0} & \mathbf{0} & I & D_{N/4} \\ \mathbf{0} & \mathbf{0} & I & -D_{N/4} \end{pmatrix} \times \begin{pmatrix} DFT_{N/4} & \mathbf{0} & \mathbf{0} \\ \mathbf{0} & DFT_{N/4} & \mathbf{0} & \mathbf{0} \\ \mathbf{0} & \mathbf{0} & DFT_{N/4} & \mathbf{0} \\ \mathbf{0} & \mathbf{0} & \mathbf{0} & DFT_{N/4} \end{pmatrix} \begin{pmatrix} \boldsymbol{x}^{(ee)} \\ \boldsymbol{x}^{(ee)} \\ \boldsymbol{x}^{(oe)} \\ \boldsymbol{x}^{(oe)} \\ \boldsymbol{x}^{(oo)} \end{pmatrix}$$

where the vectors $\mathbf{x}^{(e)}$ and $\mathbf{x}^{(o)}$ have been further split into evenand odd-indexed entries. Clearly, if this factorization is repeated, we obtain a factorization

Iterating the factorization 2

$$\mathsf{DFT}_{N} = \prod_{k=1}^{\log_{2} N} \begin{pmatrix} I & D_{N/2^{k}} & \mathbf{0} & \mathbf{0} & \cdots & \mathbf{0} & \mathbf{0} \\ I & -D_{N/2^{k}} & \mathbf{0} & \mathbf{0} & \cdots & \mathbf{0} & \mathbf{0} \\ \mathbf{0} & \mathbf{0} & I & D_{N/2^{k}} & \cdots & \mathbf{0} & \mathbf{0} \\ \mathbf{0} & \mathbf{0} & I & -D_{N/2^{k}} & \cdots & \mathbf{0} & \mathbf{0} \\ \vdots & \vdots & \vdots & \vdots & \vdots & \mathbf{0} & \mathbf{0} \\ \mathbf{0} & \mathbf{0} & \mathbf{0} & \mathbf{0} & \cdots & I & D_{N/2^{k}} \\ \mathbf{0} & \mathbf{0} & \mathbf{0} & \mathbf{0} & \cdots & I & -D_{N/2^{k}} \end{pmatrix} P.$$
(1)

FFT implementation

```
function y = FFTImpl(x, FFTKernel)
x = bitreverse(x);
y = FFTKernel(x);
```

```
function y = FFTKernelStandard(x)
N = size(x, 1);
if N == 1
    y = x;
else
    xe = FFTKernelStandard(x(1:(N/2)));
    xo = FFTKernelStandard(x((N/2+1):N));
    D = exp(-2*pi*1j*(0:(N/2-1))'/N);
    xo = xo.*D;
    y = [ xe + xo; xe - xo];
end
```

y = FFTImpl(x, @FFTKernelStandard);