## ■ 1.4 Fast Fourier Transform (FFT) Algorithm

Fast Fourier Transform, or FFT, is any algorithm for computing the $N$-point DFT with a computational complexity of $\mathcal{O}(N \log N)$. It is *not* a new transform, but simply an efficient method of calculating the DFT of $x(n)$.

If we assume that $N$ is even, we can write the $N$-point DFT of $x(n)$ as

$$\begin{aligned} X^{(N)}(k) &= \sum_{n \text{ is even: } n=2m, m=0,1,\cdots,\frac{N}{2}-1} x(n)e^{-j\frac{2\pi k}{N}n} \\ &\quad + \sum_{n \text{ is odd: } n=2l+1, l=0,1,\cdots,\frac{N}{2}-1} x(n)e^{-j\frac{2\pi k}{N}n} \\ &= \sum_{m=0}^{\frac{N}{2}-1} x(2m)e^{-j\frac{2\pi k}{N}2m} + \sum_{l=0}^{\frac{N}{2}-1} x(2l+1)e^{-j\frac{2\pi k}{N}(2l+1)} \end{aligned} \tag{1.31}$$

We make the following substitutions:

$$\begin{aligned} x_0(m) &= x(2m), \text{ where } m = 0, \cdots, \frac{N}{2}-1, \\ x_1(l) &= x(2l+1), \text{ where } l = 0, \cdots, \frac{N}{2}-1. \end{aligned}$$

Rewriting Eq. (1.31), we get

$$\begin{aligned} X^{(N)}(k) &= \sum_{m=0}^{\frac{N}{2}-1} x_0(m)e^{-j\frac{2\pi k}{\frac{N}{2}}m} + e^{-j\frac{2\pi k}{N}} \sum_{l=0}^{\frac{N}{2}-1} x_1(l)e^{-j\frac{2\pi k}{\frac{N}{2}}l} \\ &= X_0^{(\frac{N}{2})}(k) + e^{-j\frac{2\pi k}{N}} X_1^{(\frac{N}{2})}(k), \end{aligned} \tag{1.32}$$

where $X_0^{(\frac{N}{2})}(k)$ is the $\frac{N}{2}$-point DFT of the even-numbered samples of $x(n)$ and $X_1^{(\frac{N}{2})}(k)$ is the $\frac{N}{2}$-point DFT of the odd-numbered samples of $x(n)$. Note that both of them are $\frac{N}{2}$-periodic discrete-time functions.

We have the following algorithm to compute $X^{(N)}(k)$ for $k = 0, \cdots, (N-1)$ :

1. Compute $X_0^{(\frac{N}{2})}(k)$ for $k = 0, \cdots, \frac{N}{2}-1$.
2. Compute $X_1^{(\frac{N}{2})}(k)$ for $k = 0, \cdots, \frac{N}{2}-1$.
3. Perform the computation (1.32) with $N$ complex multiplications and $N$ complex additions.

Actually, it is possible to use fewer than $N$ complex multiplications. Let

$$W_N = e^{-j\frac{2\pi}{N}}.$$

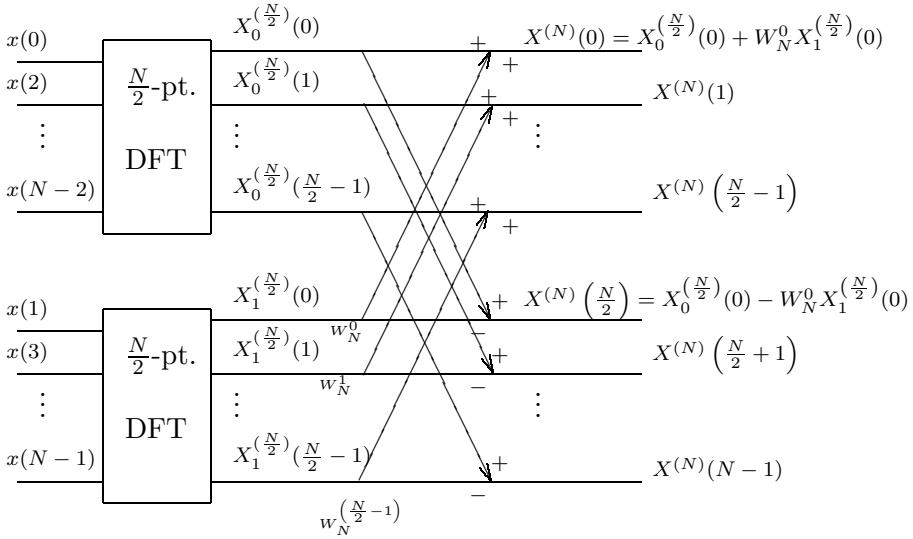


**Figure 1.36.** The FFT algorithm.

Then

$$\begin{aligned} W_N^{k+\frac{N}{2}} &= e^{-j\left(\frac{2\pi k}{N}+\pi\right)} \\ &= -e^{-j\frac{2\pi k}{N}} \\ &= -W_N^k \end{aligned}$$

Therefore,

$$\begin{aligned} X^{(N)}(k) &= X_0^{\left(\frac{N}{2}\right)}(k) + W_N^k X_1(k) \quad \text{for } k = 0, \cdots, \frac{N}{2}-1, \\ X^{(N)}\left(k+\frac{N}{2}\right) &= X_0^{\left(\frac{N}{2}\right)}(k) - W_N^k X_1(k) \quad \text{for } k = 0, \cdots, \frac{N}{2}-1, \end{aligned}$$

as illustrated in Fig. 1.36. This shows that we do not need to actually perform $N$ complex multiplications, but only $\frac{N}{2}$. [8]

Fig. 1.37 illustrates the recursive implementation of the FFT supposing that $N = 2^M$. There is a total of $M = \log_2 N$ stages of computation, each requiring $\frac{3}{2}N$ complex operations. Hence, the total computational complexity is $\mathcal{O}(N \log N)$. We see that the process ends at a 1-point DFT. A 1-point DFT is the sample of the original signal:

$$X(0) = \sum_{n=0}^{0} x(n) e^{-j\left(\frac{2\pi \cdot 0}{1}\right)n} = x(0).$$

The following remarks apply to the FFT:

[8] Actually, slightly fewer if we do not count multiplications by $\pm 1$ and $\pm j$.

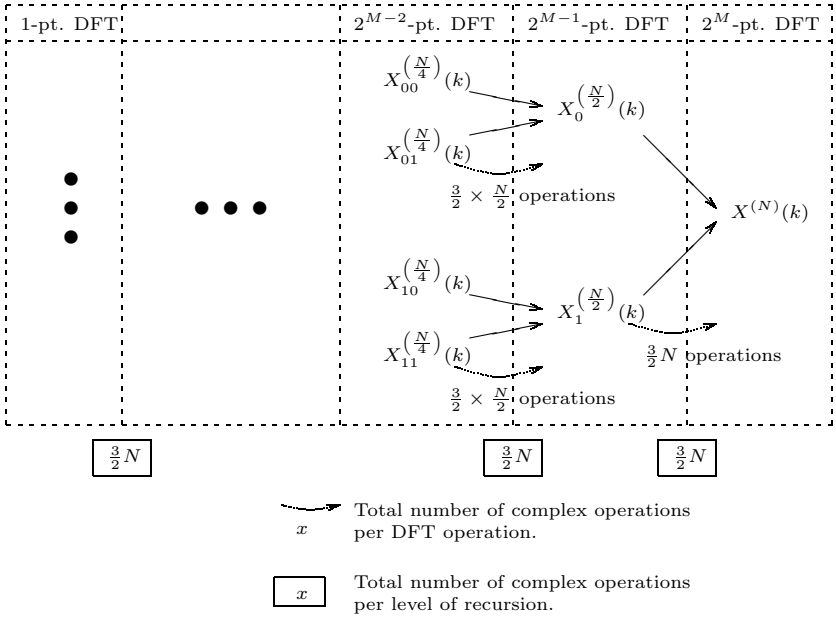


**Figure 1.37.** The recursive implementation of the FFT supposing that $N = 2^M$. There is a total of $M = \log_2 N$ stages of computation, each requiring $\frac{3}{2}N$ complex operations. Hence, the total computational complexity is $\mathcal{O}(N \log N)$.

1. For large $N$, the FFT is much faster than the direct application of the definition of DFT, which is of complexity $\mathcal{O}(N^2)$.

2. The particular implementation of the FFT described above is called *decimation-in-time radix-2 FFT*.

3. The number of operations required by an FFT algorithm can be approximated as $CN \log N$, where $C$ is a constant. There are many variations of FFT aimed at reducing this constant–e.g., if $N = 3^M$, it may be better to use a radix-3 FFT.

4. Note that

$$\begin{aligned}\left\{\frac{1}{N}\mathrm{DFT}[x^*(n)]\right\}^* &= \left\{\frac{1}{N}\sum_{n=0}^{N-1} x^*(n)e^{-j\left(\frac{2\pi k}{N}\right)n}\right\}^* \\ &= \frac{1}{N}\sum_{n=0}^{N-1} x(n)e^{j\left(\frac{2\pi k}{N}\right)n}\end{aligned}$$

   which is the IDFT of $x(n)$. Thus, the FFT can also be used to compute the IDFT.

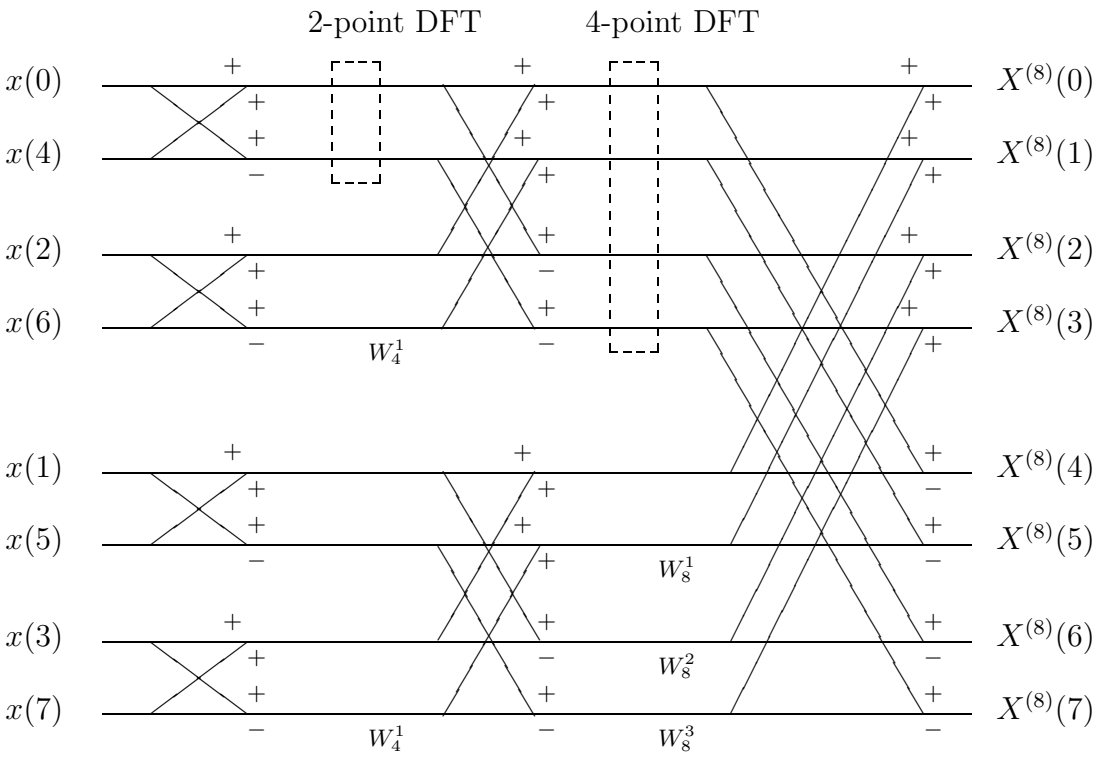


**Figure 1.38.** The 8-point FFT.

**Example 1.26.** *The 8-point FFT is depicted in Fig. 1.38. The values of the twiddle factors are:*

$$\begin{aligned} W_2 &= e^{-j\frac{2\pi}{2}} = -1, \\ W_4 &= e^{-j\frac{2\pi}{4}} = -j, \\ W_8 &= e^{-j\frac{2\pi}{8}}. \quad \blacksquare \end{aligned}$$

Recall that the DFT is a matrix multiplication (Fig. 1.35). One stage of the FFT essentially reduces the multiplication by an $N \times N$ matrix to two multiplications by $\frac{N}{2} \times \frac{N}{2}$ matrices. This reduces the number of operations required to calculate the DFT by almost a factor of two (Fig. 1.39).

Another interpretation of FFT involves analyzing the matrix

$$A_{k,L} = \frac{1}{\sqrt{2}} \begin{pmatrix} 1 & e^{-j\frac{2\pi k}{2^L}} \\ 1 & -e^{-j\frac{2\pi k}{2^L}} \end{pmatrix},$$

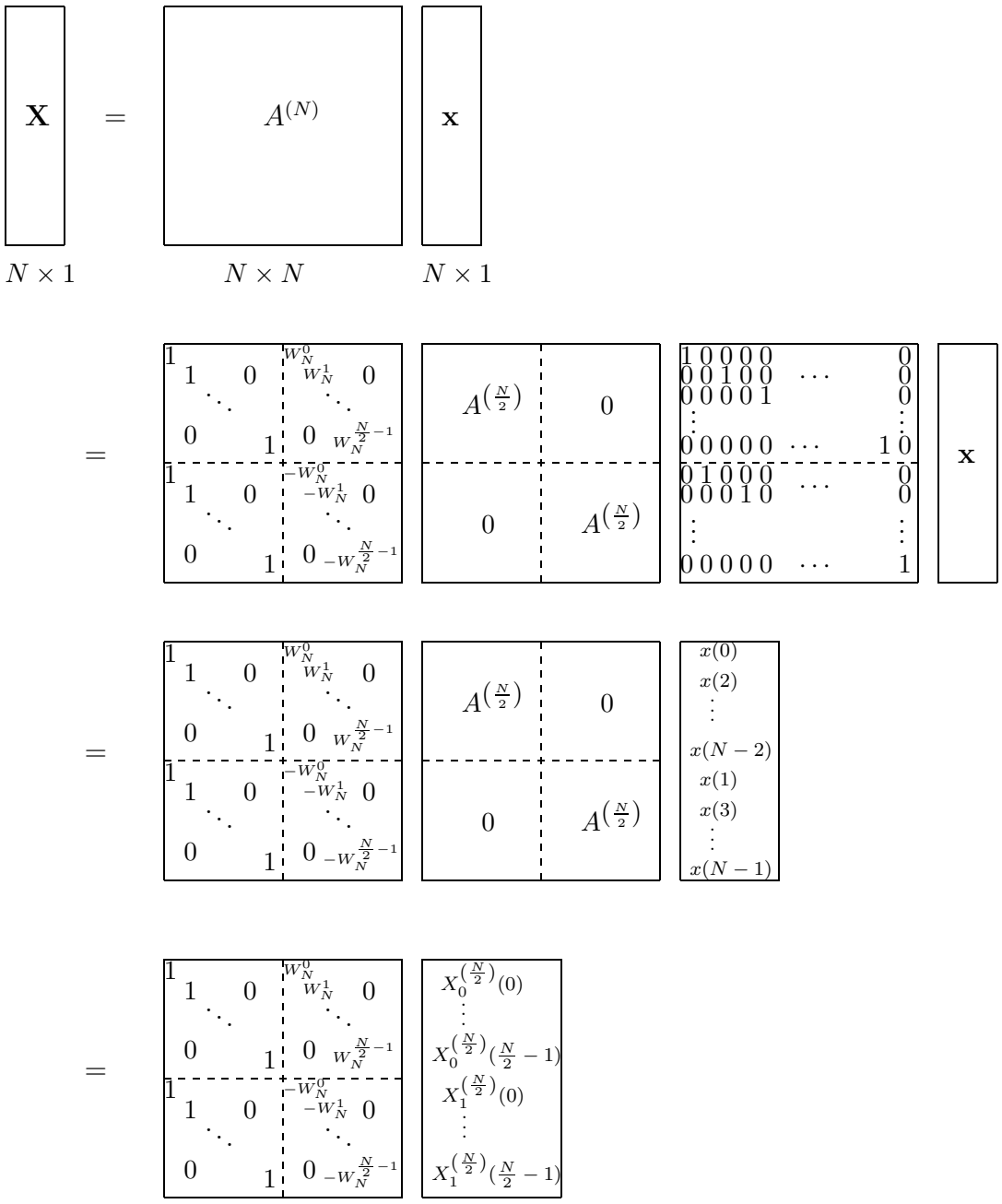


**Figure 1.39.** The FFT reduces the number of operations required to calculate the DFT by reducing $A^{(N)}$ to two $A^{\left(\frac{N}{2}\right)}$ that is only half the size of $A^{(N)}$. This operation is repeated with every recursion until we reach the 1-point DFT.

where $k$ and $L$ are nonnegative integers such that $k < 2^L$. Note that

$$\begin{aligned}
\langle A_{k,L}\mathbf{x}, A_{k,L}\mathbf{y}\rangle &= (A_{k,L}\mathbf{y})^H(A_{k,L}\mathbf{x}) \\
&= \mathbf{y}^H A_{k,L}^H A_{k,L}\mathbf{x} \\
&= \mathbf{y}^H \frac{1}{\sqrt{2}}\begin{pmatrix} 1 & 1 \\ e^{j\frac{2\pi k}{N}} & -e^{j\frac{2\pi k}{N}} \end{pmatrix} \frac{1}{\sqrt{2}}\begin{pmatrix} 1 & e^{-j\frac{2\pi k}{N}} \\ 1 & -e^{-j\frac{2\pi k}{N}} \end{pmatrix}\mathbf{x} \\
&= \mathbf{y}^H \frac{1}{2}\begin{pmatrix} 2 & 0 \\ 0 & 2 \end{pmatrix}\mathbf{x} = \mathbf{y}^H\mathbf{x} = \langle \mathbf{x}, \mathbf{y}\rangle,
\end{aligned}$$

i.e., multiplication by $A_{k,L}$ preserves distances and angles — roughly speaking, it is a rotation or reflection. Continuing the matrix decomposition of Fig. 1.39 further until we get the full FFT, it can be shown that FFT consists of $\frac{N}{2}\log N$ multiplications by $2\times 2$ matrices of the form $\sqrt{2}A_{k,L}$, each operating on a pair of coordinates.[9] Therefore, FFT breaks down the multiplication by the DFT matrix $A$ into elementary planar transformations.

## ■ 1.4.1 Fast Computation of Convolution

Consider a linear system described by

$$\mathbf{y} = S\mathbf{x}, \tag{1.33}$$

where $\mathbf{x}$ is the $N\times 1$ input vector, representing an $N$-periodic input signal; $S$ is an $N\times N$ matrix; and $\mathbf{y}$ is the $N\times 1$ output vector, representing an $N$-periodic output signal. What conditions must the matrix $S$ satisfy in order for the system to be time-invariant, i.e., invariant to circular shifts of the input vector?

Note that a circular shift by one sample is

$$\begin{pmatrix} x(0) \\ x(1) \\ x(2) \\ \vdots \\ x(N-1) \end{pmatrix} \rightarrow \begin{pmatrix} x(-1) = x(N-1) \\ x(0) \\ x(1) \\ \vdots \\ x(N-2) \end{pmatrix}.$$

Let the first column of $S$ be

$$\mathbf{h} = \begin{pmatrix} h(0) \\ h(1) \\ h(2) \\ \vdots \\ h(N-1) \end{pmatrix}.$$

[9]The same conclusion can be reached by examining an FFT diagram such as Fig. 1.38.

Note that when

$$\mathbf{x} = \begin{pmatrix} 1 \\ 0 \\ 0 \\ \vdots \\ 0 \end{pmatrix}, \text{ then } \mathbf{y} = \mathbf{h},$$

and when

$$\mathbf{x} = \begin{pmatrix} 0 \\ 1 \\ 0 \\ \vdots \\ 0 \end{pmatrix},$$

then $\mathbf{y}$ is the second column of $S$, which therefore, in order for $S$ to be invariant to circular shifts, must be equal to:

$$\begin{pmatrix} h(N-1) \\ h(0) \\ h(1) \\ \vdots \\ h(N-2) \end{pmatrix}.$$

Similarly, when

$$\mathbf{x} = \begin{pmatrix} 0 \\ 0 \\ 1 \\ 0 \\ \vdots \\ 0, \end{pmatrix}, \text{ then } \mathbf{y} \text{ is the third column of } S, \text{ etc.}$$

Thus, the matrix $S$ must have the following structure:

$$S = \begin{pmatrix} h(0) & h(N-1) & h(N-2) & \cdots & h(1) \\ h(1) & h(0) & h(N-1) & \cdots & h(2) \\ h(2) & h(1) & h(0) & \cdots & h(3) \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ h(N-1) & h(N-2) & h(N-3) & \cdots & h(0) \end{pmatrix}.$$

This is called a *circulant matrix*. We can then write Eq. (1.33) as

$$\begin{aligned} y(n) &= \sum_{m=0}^{N-1} x(m)h(n-m) \\ &= \sum_{m=0}^{N-1} x(m)h((n-m) \bmod N) \qquad (1.34) \\ &= x \circledast h(n) = x \mathbin{\textcircled{N}} h \qquad (1.35) \end{aligned}$$

Eq. (1.35) is called a *circular convolution* or a *periodic convolution*. Note that formula (1.34) works even when $x$ or $h$ are non-periodic. Observe the following:

- For $y(0)$, the sum of the indices of $x$ and $h$ is always 0 mod $N$ for every term.

$$y(0) = x(0)h(0) + x(1)h(N-1) + x(2)h(N-2) + \cdots + x(N-1)h(1)$$

- For $y(1)$, the sum of the indices of $x$ and $h$ is always 1 mod $N$ for every term.

$$y(1) = x(0)h(1) + x(1)h(0) + x(2)h(N-1) + \cdots + x(N-1)h(2)$$

This is true for all $y(k)$, $k = 0, 1, \cdots, N-1$.

What are the eigenvectors of $S$? Let us try

$$\mathbf{g}_k = \begin{pmatrix} \frac{1}{N} \\ \frac{1}{N} e^{j\frac{2\pi k}{N}\cdot 1} \\ \frac{1}{N} e^{j\frac{2\pi k}{N}\cdot 2} \\ \vdots \\ \frac{1}{N} e^{j\frac{2\pi k}{N}\cdot (N-1)} \end{pmatrix}, \text{ where } k = 0, 1, \cdots, N-1.$$

We have:

$$\begin{aligned} y(n) &= h(n) \circledast \mathbf{g}_k \\ &= \sum_{m=0}^{N-1} h(m) g_k(n-m) \\ &= \sum_{m=0}^{N-1} h(m) \frac{1}{N} e^{j\frac{2\pi k}{N}(n-m)} \\ &= \left\{ \sum_{m=0}^{N-1} h(m) e^{-j\frac{2\pi k}{N}m} \right\} \frac{1}{N} e^{j\frac{2\pi k}{N}n} \\ &= \underbrace{H(k)}_{\text{DFT of } h} \frac{1}{N} e^{j\frac{2\pi k}{N}n} \end{aligned}$$

Hence we have that

$$S\mathbf{g}_k = H(k)\mathbf{g}_k$$

where $\mathbf{g}_k$ is the $k$-th eigenvector and $H(k)$ gives the corresponding eigenvalue. Therefore,

$$S\underbrace{\begin{pmatrix} \mathbf{g}_0 & \mathbf{g}_1 & \cdots & \mathbf{g}_{N-1} \end{pmatrix}}_{\text{The IDFT matrix } B} = \begin{pmatrix} \mathbf{g}_0 & \mathbf{g}_1 & \cdots & \mathbf{g}_{N-1} \end{pmatrix} \begin{pmatrix} H(0) & & & 0 \\ & H(1) & & \\ & & \ddots & \\ 0 & & & H(N-1) \end{pmatrix}.$$

Then $S$ can be written as:

$$S = B \begin{pmatrix} H(0) & & & 0 \\ & H(1) & & \\ & & \ddots & \\ 0 & & & H(N-1) \end{pmatrix} A,$$

where the DFT matrix $A$ is:

$$A = NB^H = \begin{pmatrix} \mathbf{g}_0^H \\ \mathbf{g}_1^H \\ \vdots \\ \mathbf{g}_{N-1}^H \end{pmatrix}.$$

Complex exponentials are the eigenvectors of circulant matrices. They *diagonalize* circulant matrices. Thus, for any $\mathbf{x} \in \mathbb{C}^N$,

$$S\mathbf{x} = B \begin{pmatrix} H(0) & & & 0 \\ & H(1) & & \\ & & \ddots & \\ 0 & & & H(N-1) \end{pmatrix} A\mathbf{x}.$$

Let us compare two algorithms for computing the circular convolution of $\mathbf{x}$ and $\mathbf{h}$.

**Algorithm 1** Directly perform the multiplication $S\mathbf{x}$. This has computational complexity $\mathcal{O}\left(N^2\right)$.

**Algorithm 2** 1. Represent $\mathbf{x}$ in the eigenbasis of $S$, i.e., the Fourier basis,

$$\mathbf{X} = A\mathbf{x}.$$

This step can be done with FFT whose complexity is $\mathcal{O}(N \log N)$.

| Step 1 | Step 2 | Step 3 |
|---|---|---|
| $x(n) \overset{\text{N-point DFT}}{\longrightarrow} X(k)$<br>$h(n) \overset{\text{N-point DFT}}{\longrightarrow} H(k)$ | $Y(k) = X(k)H(k)$ | $Y(k) \overset{\text{N-point IDFT}}{\longrightarrow} y(n) = x \circledast h(n)$ |

**Figure 1.40.** An illustration of the FFT implementation of the circular convolution.

2. Compute the representation of $\mathbf{y}$ in the eigenbasis of $S$:

$$\mathbf{Y} = \begin{pmatrix} H(0) & & & 0 \\ & H(1) & & \\ & & \ddots & \\ 0 & & & H(N-1) \end{pmatrix} \mathbf{X}.$$

   This computation has complexity $\mathcal{O}(N)$.

3. Reconstruct $\mathbf{y}$ from its Fourier coefficients:

$$\mathbf{y} = B\mathbf{Y}.$$

   This has complexity $\mathcal{O}(N \log N)$, if done using the FFT.

This algorithm is summarized in Fig. 1.40. Its total complexity is $\mathcal{O}(N \log N)$.

(Note that the second algorithm does not necessarily perform better for *any* matrix.)

**Example 1.27.** *This example explores the relationship between the convolution and the circular convolution. Let $x$ and $h$ be $N$-periodic signals, and let*

$$\begin{aligned} x_z &= \begin{cases} x(n), & 0 \le n \le N-1 \\ 0, & \textit{otherwise} \end{cases} \\ h_z &= \begin{cases} h(n), & 0 \le n \le N-1 \\ 0, & \textit{otherwise} \end{cases} \end{aligned}$$

*If we let*

$$\begin{aligned} y_z(n) &= x_z * h_z(n) \\ y(n) &= x \circledast h(n) \end{aligned}$$

*then $y(n)$ can be expressed as*

$$y(n) = \begin{cases} y_z(n) + y_z(N+n), & n = 0, 1, \cdots, N-2 \\ y(N-1), & n = N-1 \end{cases}$$

*Note that the overlap of $y_z(n)$ and $y_z(N+n)$ causes temporal aliasing in the resulting $y(n)$. This is the main difference between convolution and circular convolution.*

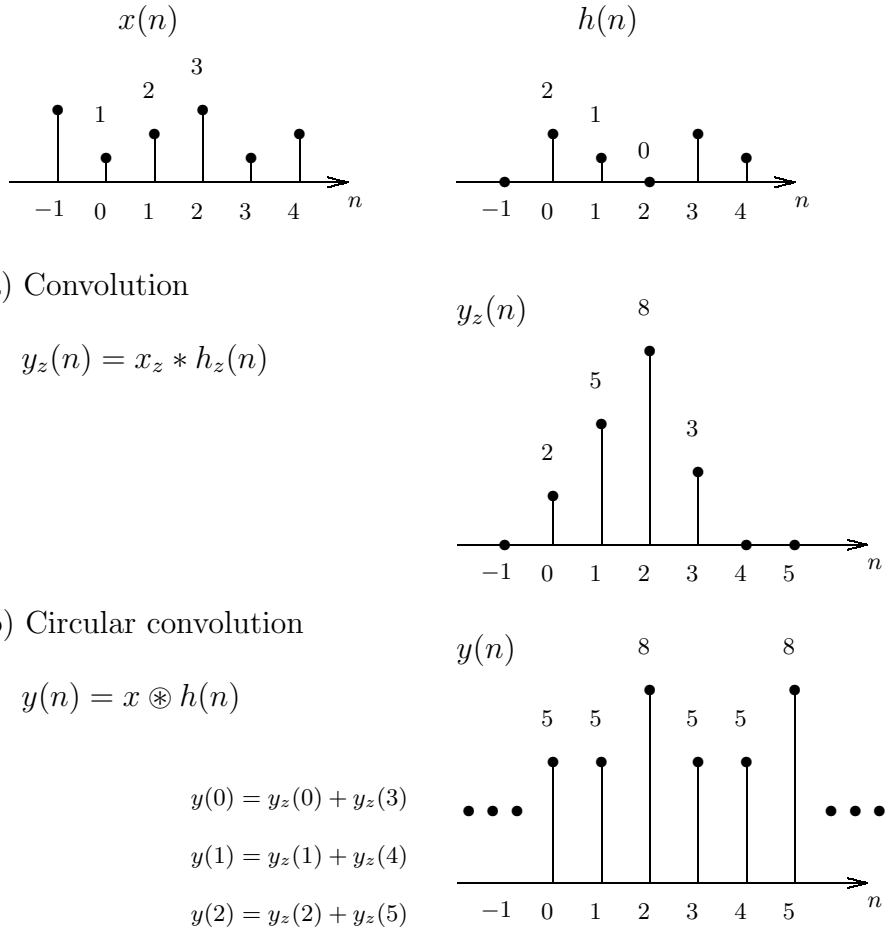


**Figure 1.41.** A comparison between circular convolution and convolution.

*Fig. 1.41 illustrates the effect of temporal aliasing. To remove or minimize the effect of temporal aliasing, we could zero-pad $x$ and $h$ so that the temporal replicas are spread further apart, and thus, overlapping would not occur.*