NTT 201 - Foundations of NTT Hardware Design

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Chapter 1

General Theory

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1.1 Introduction

NTT, or Number Theoretic Transform. is the term used to describe a Discrete Fourier Transform (DFT) over finite fields. In general, there's nothing unique about NTT that does not apply to DFT and visa-versa. However, the uses of NTT, its typical operating point, and some subtleties make it a very different beast. This note is twofold. In this chapter, we describe NTT from the use-case point-of-view, emphasizing what it is good for and what it is not. In the next chapter, we show how an NTT of a specific operating point can be optimally implemented in hardware.

1.2 From FT to DFT

1.2.1 FT

Fourier Transform (FT) [1] converts a continuous signal¹ from the time domain to the frequency domain and is defined as

$$X(\varepsilon) = \int_{-\infty}^{\infty} x(t)e^{-i2\pi\varepsilon t}dt$$
 (1.1)

FT exists under complicated necessary conditions, but a generally sufficient condition is that $x(t) \in \mathcal{L}_1$ where \mathcal{L}_1 is the set of all absolutely integrable functions.² Other than that, x(t) is an infinite, non-periodic function of time. FT is a unitary linear transformation and obeys Parseval's theorem [2]. One can think of $X(\varepsilon_0)$ as the quantity of signal energy at frequency ε_0 since it is exactly the correlation between the signal and the discrete tone $e^{-i2\pi\varepsilon_0 t}$.

¹We tend to refer to the time-domain function as a signal in the context of FT.

²a.k.a. Lebesgue integrable functions.

1.2.2 DTFT

The Discrete-Time FT (DTFT) [3] is an FT for discrete-time signals and is defined as

$$X(f) = \sum_{-\infty}^{\infty} x[n]e^{-i2\pi fn}$$
(1.2)

The sampled discrete tone $e^{-i2\pi fn}$ is periodic, leading to a periodic frequency-domain function X(f) with maximum frequency at f=0.5. The DTFT can be thought of as a folded version of FT where the domain $\varepsilon \in (-\infty, \infty)$ is folded onto $f \in (-0.5, 0.5]$. DTFT can still be thought of in the sense of signal energy per frequency, but now the frequency f contains the accumulated energy of $X(\varepsilon)$ at frequencies $\{..., \varepsilon-2, \varepsilon-1, \varepsilon, \varepsilon+1, \varepsilon+2, ...\}$. The folding phenomenon in the frequency domain, often referred to as aliasing, is associated with time-domain sampling. In that sense, we can think of x[n] as a sampled version of x(t) (i.e. x[n] = x(nT) where T is the sampling period).

1.2.3 DFT

But what about periodic signals? On the one hand, a periodic signal has infinite energy. On the other hand, a single period is deterministically sufficient to represent it. So how about a transformation that represents a single period of a periodic signal? As it turns out, this is possible. For continuous-time periodic signals, this is called a Fourier Series (FS) [4]. For discrete-time periodic signals, it is called a DFT [5] and is defined as

$$X[k] = \sum_{n=0}^{N-1} x[n]e^{-i2\pi \frac{k}{N}n}$$
(1.3)

The frequency equivalent for DFT is $\frac{k}{N}$ for $k \in [0, N-1]$. DFT can be viewed as a normalized DTFT of a periodic signal. It is for that reason that for DFT we typically refer to signal power rather than energy.

The following table summarises the relationship between different transforms:

	FT	DTFT	FS	DFT
Periodic TD (Sampled FD)			√	✓
Periodic FD (Sampled TD)		✓		√

Note that it may seem that DFT is finite-time and finite-frequency. In fact, that is not the case but it is rather periodic in both time and frequency. This, as we shall see, has much influence on some of its main properties.

1.3 Properties of DFT

The set of discrete tones used in the DFT can be rewritten as

$$e^{-i2\pi \frac{k}{N}n} = \left(e^{-i\frac{2\pi}{N}}\right)^{kn} = \omega^{kn} \tag{1.4}$$

³An exact comparison between FT and DTFT requires normalization by the sampling period $T = f^{-1}$.

where ω is termed the N'th root-of-unity, since N is the smallest integer such that $\omega^N = 1$. We can now rewrite the DFT in its generalized form

$$X[k] = \sum_{n=0}^{N-1} x[n]\omega^{kn}$$
 (1.5)

where ω^{kn} is a generalized discrete tone.

Replacing x[n] with coefficients x_n of and substituting z for ω^k , results in the z-transform of the sequence $\{x_n\}_{n=0}^{N-1}$

$$X(z) = \sum_{n=0}^{N-1} x_n z^n \tag{1.6}$$

which is simply a polynomial of order N-1 in z.⁴ With that, we can think of the transform X[k] as polynomial evaluations of X(z) over the root-of-unity powers⁵ (i.e. $X[k] = X(\omega^k)$). This perspective is extremely useful in understanding some of the most useful properties of DFT.

1.3.1 Convolution

Linear convolution of two length-N sequences f_n and g_n is defined as

$$(f * g)_n = \sum_{m=0}^{N-1} f_m g_{n-m}$$
(1.7)

where both sequences are set zero for all $n \notin [0, N-1]$. In the polynomial form, this is described as

$$(F \cdot G)(z) = F(z)G(z) \tag{1.8}$$

$$= \sum_{m=0}^{N-1} f_m z^m \cdot \sum_{k=0}^{N-1} g_k z^k \tag{1.9}$$

$$=\sum_{m=0}^{N-1}\sum_{k=0}^{N-1}f_m \cdot g_k z^{m+k}$$
 (1.10)

$$=\sum_{m=0}^{N-1}\sum_{n=m}^{m+N-1}f_m\cdot g_{n-m}z^n$$
(1.11)

$$= \sum_{n=0}^{2N-1} \left(\sum_{m=0}^{N-1} f_m \cdot g_{n-m} \right) z^n \tag{1.12}$$

where equality (1.12) is obtained by using the previous definition that $g_n \equiv 0$ for all $n \notin [0, N-1]$. Note that the term in parenthesis is equivalent to our previous definition of linear convolution (1.7).

⁴The Region of Convergence (RoC) is obviously $z \leq 1$ so it consists of $[\omega]$

⁵Note how the root-of-unity powers are an equidistant partition of the unit circle.

So how can we use this? Imagine you have two length-N sequences and you would like to calculate their linear convolution of length 2N-1, or you have coefficients of two degree-(N-1) univariate polynomials and you want to calculate the coefficients of their product polynomial of degree-(2N-2). The above claims that the point-wise multiplication of two polynomials' evaluations results in the product polynomial. The nice thing is that the product of evaluations can be used to calculate the corresponding coefficients by Inverse DFT (IDFT).⁶ This is nearly sufficient and only missing one crucial ingredient. This ingredient is an outcome of the Fundamental Theorem of Algebra [6] which states that any polynomial of degree-(L-1) is uniquely defined by any L distinct evaluations. This means that any 2N-1 evaluations of the product polynomial $(F \cdot G)(z)$ suffice to uniquely define the coefficients $(f * g)_n$. With that, let us define the following convolution algorithm for two sequences of length-N

$$(f * g)_{n=0}^{2N-1} = IDFT_{2N-1} (DFT_{2N-1}(f_n) \circ DFT_{2N-1}(g_n))$$
(1.13)

where \circ denotes the Hadamard element-wise product, subscript \square_{2N-1} depicts a transform's length, and all sequences are assumed to be zero-padded to the required length as needed. The advantage of the r.h.s. of (1.13), as discussed later in this work, is that its typical computational complexity is $\mathcal{O}(N \log N)$, whereas the complexity of the naive linear convolution is $\mathcal{O}(N^2)$.

Finally, one could ask what would result from the following algorithm with all transforms shortened to length N

$$(f \circledast g)_{n=0}^{N-1} = IDFT_N (DFT_N(f_n) \circ DFT_N(g_n))$$
(1.14)

As it turns out, and somewhat hinted by the l.h.s. notation of (1.14), the above is an algorithm for calculating the cyclic convolution of two length-N sequences or the polynomial product of two polynomials in a polynomial ring $P[z]/(z^N-1)$. To see why this is so let us reexamine the polynomial product in the context of the ring $P[z]/(z^N-1)$ (i.e. $F(z)G(z)/(z^N-1)$). Clearly, the modulo restriction adds the cyclic constraint $z^{k+N}=z^k$ which when plugged into (1.12) leads to the following

$$F(z)G(z)/(z^{N}-1) = \sum_{n=0}^{2N-1} \left(\sum_{m=0}^{N-1} f_m \cdot g_{n-m}\right) z^n$$
(1.15)

$$= \sum_{n=0}^{N-1} \left(\sum_{m=0}^{N-1} f_m \cdot g_{n-m} \right) z^n + \sum_{n=N}^{2N-1} \left(\sum_{m=0}^{N-1} f_m \cdot g_{n-m} \right) z^n$$
 (1.16)

$$= \sum_{n=0}^{N-1} \left(\sum_{m=0}^{N-1} f_m \cdot g_{n-m} \right) z^n + \left(\sum_{m=0}^{N-1} f_m \cdot g_{n-m+N} \right) z^n$$
 (1.17)

$$= \sum_{n=0}^{N-1} \left(\sum_{m=0}^{N-1} f_m \cdot (g_{n-m} + g_{n-m+N}) \right) z^n$$
 (1.18)

$$= \sum_{n=0}^{N-1} \left(\sum_{m=0}^{N-1} f_m \cdot \widetilde{g}_{n-m} \right) z^n \tag{1.19}$$

⁶DFT is a unitary transformation and is thus bijective (i.e. invertible). This property is discussed hereafter.

where \tilde{g}_n is the N-periodic concatenation of g_n . The resulting polynomial (1.19) is of maximum degree N-1 and is thus uniquely defined by N distinct evaluations of the product F(z)G(z) which immediately leads to algorithm (1.14).

1.3.2Interpolation

Interpolation is the process of calculating the value of an evaluation of a polynomial for some point in its domain. Since the polynomial coefficients (or time domain signal) are well-defined, the interpolated value is unique. For a single sample, this can be achieved by directly evaluating the polynomial from its coefficients or by using a form of Lagrange Interpolation such as the Barycentric Formula [7] over its available evaluations. In both cases, this is achievable with computational complexity $\mathcal{O}(N)$. It is often the case in data processing that we are interested in interpolating $L \geqslant N$ points located on a regularlyspaced grid. The naive implementation would require $\mathcal{O}(LN)$ complexity, but as it turns out, this can be performed by simple manipulation of the DFT, lowering the complexity to at most $\mathcal{O}(L \log L)$.

Suppose we have a sequence f_n of length N. One can easily use (1.6) to evaluate F(z)anywhere in its domain. With a DFT of length N one can evaluate $\Omega = \{F(z) : z \in [\omega]\}$ where ω is the N'th root-of-unity and $[\omega]$ is the cyclic multiplicative subgroup generated by ω . With a DFT of length L > N one can evaluate $\Upsilon = \{F(z) : z \in [v]\}$ where v is the L'th root-of-unity. One may ask how a length-L DFT can be performed over the sequence f_n of length N. Since we treat the time domain sequence as polynomial coefficients, the only requirement is to zero-pad it to length-L and proceed as if it were a length-L sequence. This is equivalent to an order-L-1 univariate polynomial whose L-N most significant monomials are zero. This leads to the following algorithm for arbitrary interpolation from subdomain $[\omega] \to [v]$

$$F([v]) = DFT_L(IDFT_N(F([\omega])))$$
(1.20)

where $F([\omega]) \equiv \{F(z) : z \in [\omega]\}$ and zero-padding is assumed implicitly.

When $L = \lambda N$ for some $\lambda \in \mathbb{N}$ then $[\omega] = [v^{\lambda}] \leq [v]$ which means that

$$[v] = \bigcup_{l=0}^{\lambda-1} v^l[\omega] \tag{1.21}$$

where $v^l[\omega]$ are cosets of $[\omega]$. The key observation is that $F([\omega]) = F([v^{\lambda}]) \subset F([v])$ and it seems that in algorithm (1.20) the evaluations $F([\omega])$ would appear in both the input and output (i.e. some computational effort would be spent on recomputing $F([\omega])$). Can we somehow avoid this unnecessary effort? As it turns out, we can and this is particularly beneficial when λ is small (e.g. $\lambda = 2$ as is often the case). To see how this can be done, let us plug the k'th element from coset $v^{l}[\omega]$ into (1.6)

$$F(v^l \omega^k) = \sum_{n=0}^{N-1} f_n(v^l \omega^k)^n$$
(1.22)

$$=\sum_{n=0}^{N-1} v^{ln} f_n \omega^{kn} \tag{1.23}$$

In the context of the transform of length L (i.e. with v as the root of unity) the above is exactly the formula for $F[\lambda k + l]$, the evaluations of coset l. This leads to the following algorithm for evaluating coset l from an original sequence of N evaluations

$$F(v^{l}[\omega]) = DFT_{N}([v^{l}] \circ IDFT_{N}(F([\omega])))$$
(1.24)

where the Hadamard multiplication by $[v^l]$ is called modulation. Note how modulation of the coefficients by $[v^l]$ translates to a shift of the evaluations by v^l (i.e. $F([\omega]) \to F(v^l[\omega])$). As an example, for $\lambda = 2$ the cost of interpolation reduces from a length-N IDFT and a length-N DFT to two length-N transforms.

Note that we used the frequency-shift/time-modulation property. The dual of this property (i.e. time-shift/frequency-modulation) holds in very much the same way and is extremely common in signal processing applications.

1.4 Fast DFT (FFT)

Fast Fourier Transform (FFT) is the common terminology for efficient methods for calculating the DFT. There are two unique methods.

The less common method, that can be shown to achieve near-linear complexity, is due to Winograd [8]. The problem with the Winograd method is that it is difficult to construct for arbitrary sizes. Constructions are known for fairly small-size transforms and as a result, it is not widely used. Although the Winograd FFT is utilized in Part 2 of this work, we will not explore it further here.

The common method for constructing FFT is typically associated with a 1965 paper due to Cooley and Tukey [9] even though it incorporates ideas that date back to Gauss as early as 1805 [10]. The method is often referred to as CT-FFT. The general idea is that when the transform length is a composite integer, it can be partitioned into smaller transforms whose results are combined to provide the desired calculation. Imagine we are interested in a length-N transformation where N = LM with $L, M \in \mathbb{N}$. With CT-FFT the complexity can be reduced from $\mathcal{O}(N^2) = \mathcal{O}(L^2M^2)$ to approximately $\mathcal{O}(LM^2 + ML^2)$. The recursive application of the above method reduces the overall complexity to $\mathcal{O}(N \log N)$.

There are two main methods to construct recursive partitioning, Decimation-In-Time (DIT) and Decimation-In-Frequency (DIF). Both methods are essentially one and it can be shown that they both produce identical results up to permutation. Starting with the following DFT for a length-N sequence f_n

$$F(\omega^k) = \sum_{n=0}^{N-1} f_n \omega^{kn}$$
(1.25)

we can construct a DIT by partitioning the sequence to L sets $\left\{ \{f_{mL+l}\}_{m=0}^{M-1} \right\}_{l=0}^{L-1}$ and proceeding to calculate the partial sums of (1.25) accordingly.⁷

$$F(\omega^k) = \sum_{l=0}^{L-1} \sum_{m=0}^{M-1} f_{mL+l} \omega^{k(mL+l)}$$
(1.26)

⁷The interleaved manner of the inner sets in the partition is often referred to as reverse-bit-ordering for DFTs of lengths that are powers of two.

1.5. NTT

$$= \sum_{l=0}^{L-1} \omega^{kl} \sum_{m=0}^{M-1} f_{mL+l}(\omega^L)^{km}$$
 (1.27)

$$= \sum_{l=0}^{L-1} (\omega^M)^{ql} \omega^{\rho l} \sum_{m=0}^{M-1} f_{mL+l}(\omega^L)^{\rho m}$$
 (1.28)

$$= \sum_{l=0}^{L-1} \eta_L^{ql} \omega^{\rho l} \sum_{m=0}^{M-1} f_{mL+l} \cdot \eta_M^{\rho m}$$
 (1.29)

$$= \sum_{l=0}^{L-1} \omega^{\rho l} \cdot \left(\sum_{m=0}^{M-1} f_{mL+l} \cdot \eta_M^{\rho m} \right) \eta_L^{ql}$$
 (1.30)

where $\eta_L = \omega^M$ and $\eta_M = \omega^L$ are the *L*-th and *M*-th roots of unity respectively and $k = qM + \rho$ where $\rho < M$. Note that the internal sum in parenthesis represents *L* length-*M* DFTs. The outputs of these DFTs are multiplied element-wise by the twiddle-factors $\omega^{\rho l}$ and the products are fed into *M* length-*L* DFTs. An intuitive way to think about this algorithm is as follows:

- 1. Organise the sequence f_n of length N in an $L \times M$ matrix, where the construction is done column-wise.
- 2. Perform an independent length-M DFT per each row of the matrix.
- 3. Multiply all matrix elements such that element (l, ρ) is multiplied by the twiddle-factor $\omega^{\rho l}$.
- 4. Perform an independent length-L DFT per each column of the matrix.

The DIF alternative is to partition the transform $\{F(\omega^k)\}_{k=0}^{N-1}$ to $\{\{F(\omega^{mL+l})\}_{m=0}^{M-1}\}_{l=0}^{L-1}$ and proceed to calculate the partial transforms. Recursing the FFT partitioning is easy and will not be discussed here. As mentioned above, CT-FFT partitioning only relies on the ability to factorize N to integral factors. As such, the finest partition is limited by the prime factors of N.⁸ It is worth noting that DFTs with lengths that are powers of two are extremely popular in practice.

1.5 NTT

NTT is a DFT over a finite-field of prime size and is typically used in cryptography. Although one can give a notion of frequency and power to the transform, it is typically used as a mathematical tool for efficiently operating with polynomials in polynomial rings. An NTT for a sequence will always exist, provided a root-of-unity of appropriate order exists. An interesting property of finite fields is that the elements of its multiplicative group all lie on the unit circle and are all roots of unity.

A question that often arises is, when is NTT a beneficial tool? In general, NTT should be used for efficiently performing convolutions and for interpolating many points that lie on regularly-spaced grids. As a rule of thumb, one should be concerned when the number of output values being calculated is smaller than the size of the NTTs being used. In those

⁸This is not the case for Winograd FFT.

cases, there are usually more efficient methods for calculation that often do not require NTT at all.

1.5.1 Example: Groth16

Groth16 [11] is a Zero Knowledge Proving (ZKP) system [12]. As part of the system, the prover is required to calculate the following quotient

$$Q(x) = \frac{A(x)B(x) - C(x)}{x^N - 1}$$
(1.31)

where A(x), B(x), and C(x) are of order N-1, and the denominator factorizes as

$$x^{N} - 1 = \prod_{n=0}^{N-1} (x - \omega^{n})$$
 (1.32)

where ω is an N'th root of unity. When the prover is honest, x^N-1 divides the nominator, and Q(x) is a polynomial of order N-1. The prover begins with the following evaluation sequences $A([\omega])$, $B([\omega])$, and $C([\omega])$. Since $[\omega]$ are all roots of both the nominator and denominator, simply plugging those evaluations in (1.31) would result in $\frac{0}{0}$ for evaluations of $Q([\omega])$ which is practically useless.

We may deduce that since the nominator is of maximum degree 2N-2, we would be required to interpolate A(x), B(x), and C(x) to at least 2N-1 distinct evaluations before proceeding. As it turns out, we can do a little better by recognizing that the result polynomial Q(x) is of order N-1. This means that N distinct evaluations suffice to represent it. An efficient method to get these N evaluations is by sampling all polynomials in N non-zero locations. This sampling is achievable by algorithm (1.24) using two length-N NTTs per source polynomial. Altogether the complexity is six length-N NTTs and some linear-time processing.

Chapter 2

Complex NTT

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2.1 Introduction

The main difference between NTT and DFT is the field over which they are defined. NTT is typically defined over a finite field of prime size (i.e. \mathbb{F}_p where p is prime), while DFT is defined over the Complex field \mathbb{C} . The reason that DFT is defined over \mathbb{C} is that the only root-of-unity over the Real field \mathbb{R} is -1 with order 2. Extraordinarily, over \mathbb{C} , a binary extension of \mathbb{R} , roots-of-unity exist for any order.

Lagrange's Theorem [13] states that for any subgroup H of a finite group G, the order of the subgroup divides the order of the group. Consequently, the order of any root-of-unity in \mathbb{F}_p , must divide p-1. In this chapter, let us limit our examination for \mathbb{F}_p such that -1 is its only root-of-unity, similarly to \mathbb{R} . In that case, $\frac{p-1}{2}$ is odd or equivalently $p=3 \mod 4$. For this case a native NTT of size a power-of-two does not exist. Similarly to the case with \mathbb{R} , this is elegantly resolved by moving to the second extension of \mathbb{F}_p . The interesting cases arise when attempting to use the Complex NTT for Real-valued sequences.

A partial treatment of this non-native NTT case was performed by Haböck *et al.* in [14]. This chapter aims to provide a simplified and more complete treatment to the subject while adding explicit constructions along with their efficient processing methods.

2.2 The Complex Field of Characteristic p

Let us define the following polynomial ring over \mathbb{F}_p

$$\mathbb{F}_p[x]/x^2 + 1 \tag{2.1}$$

When -1 is not a quadratic residue in \mathbb{F}_p (i.e. $x^2 \neq -1 \quad \forall x \in \mathbb{F}_p$), the above ring is a field of size p^2 and we will refer to it as \mathbb{C}_p , the complex field of characteristic p. Note that by Largrange's Theorem, since the size of the multiplicative group \mathbb{F}_p^* is p-1, -1 is not a quadratic residue iff $\frac{p-1}{2}$ is odd.

Defining $i = \sqrt{-1}$ in the usual way, we can now represent a field element in \mathbb{C}_p as

$$z = x + iy (2.2)$$

where $x, y \in \mathbb{F}_p$. It is beneficial to define the conjugate of z as

$$\bar{z} = x - iy \tag{2.3}$$

leading to the usual relation

$$z\bar{z} = x^2 + y^2 \tag{2.4}$$

The structure of the complex field leads to many potential operational optimizations like Karatsuba multiplication [15] and special inversion techniques [16].

Interestingly, it turns out that the unit-circle set $\{z: z\bar{z}=1\}$ is a multiplicative subgroup of \mathbb{C}_p . We will refer to this subgroup as

$$\mathbb{C}_p^{\circ} \equiv \{ z : z\bar{z} = 1 \} \tag{2.5}$$

and leave it to the reader to verify that it is indeed a subgroup. Since

$$z\bar{z} = (x+iy)(x-iy) = (x+iy)(x+iy)^p = (x+iy)^{p+1} = 1$$
 (2.6)

there are p+1 unique field elements in \mathbb{C}_p° , hence this is its size (see [17]).

2.3 Complex NTT

The obvious question is: What is the advantage in extending to Complex? As we have seen before, \mathbb{F}_p for p's such that \mathbb{C}_p exists have a single, order-2 root-of-unity (equivalently $\frac{p-1}{2}$ is odd). Since DFT sizes are based on the root-of-unity subgroup they employ, and those subgroups do not exist for sizes that are a power-of-two, efficient techniques for power-of-two FFTs are prohibitive for these cases. This is particularly the case for Mersenne primes [18]. As mentioned above, extending to \mathbb{C}_p provides root-of-unity orders up to p+1. For Mersenne primes, since p+1 is a power-of-two, this facilitates power-of-two FFTs up to size p+1.

The standard Complex NTT (CNTT) uses an N-th root-of-unity $\omega \in \mathbb{C}_p^{\circ}$ and is defined in the usual way as

$$X(\omega^k) = \sum_{n=0}^{N-1} x_n \omega^{kn}$$
(2.7)

where x_n are coefficients of the z-polynomial

$$X(z) = \sum_{n=0}^{N-1} x_n z^n$$
 (2.8)

and the transform is its evaluations $X(z = \omega^k)$ for $k \in [N]$.

2.4 Real-valued CNTT (RNTT)

Since \mathbb{F}_p is a subfield of \mathbb{C}_p , we can use the inverse of (2.7) to calculate an ICNTT for a Real-valued evaluations sequence $\{X(\omega^k)\}_{k=0}^{N-1} \in \mathbb{F}_p^N$. In what follows, we will refer to this Real-valued CNTT as RNTT. Note that for RNTT, we always assume that the evaluations are Real and the coefficients are Complex, although this can always be reversed by considering the conjugate root-of-unity.

2.4.1 Properties of RNTT

Since we assume that the evaluations are Real, it's convenient to develop the IRNTT and deduce RNTT from it. Let us start by reformulating the IRNTT as

$$x[n] = \frac{1}{N} \sum_{k=0}^{N-1} X[k] \omega^{-kn}$$
 (2.9)

where $x[n] \equiv x_n$ and $X[k] \equiv X(\omega^k)$. Since X[k] is Real, its entropy¹ matches that of a length-N vector of elements in \mathbb{F}_p . It therefore makes sense that the entropy of x[n] will remain unchanged, (since RNTT is a unitary transform). This can be exhibited using the basic conjugate-symmetry property of RNTT

$$x^*[n] = \frac{1}{N} \sum_{k=0}^{N-1} X[k] \omega^{kn} = \frac{1}{N} \sum_{k=0}^{N-1} X[k] \omega^{-k(N-n)} = x[N-n]$$
 (2.10)

where * denotes Complex-conjugation. Careful examination of this symmetry shows that $x^*[0] = x[0]$ and $x^*[N/2] = x[N/2]$ (i.e. they are Real). The remaining N-2 evaluations all have symmetric pairs (e.g. $x^*[1] = x[N-1]$). This means that x[n] can be represented compactly by 2 Real values and $\frac{N-2}{2}$ Complex values, which adds up to N scalars, as expected.²

Finally, RNTT is characterized by Real evaluations and conjugate-symmetric coefficients. Additionally, it is often convenient to think of the coefficients domain as a periodic function of n. More on this in the next section.

2.4.2 The RNTT Polynomial

Let us restate (2.8) with its explicit periodicity in the coefficients domain.

$$X(z) = \sum_{m=0}^{N-1} x_m z^m \mod z^N - 1$$
 (2.11)

where $X[k] = X(z = \omega^k)$ and $x_m = x[n = m]$.

In what follows we will prefer to present (2.11) centered around m = 0. This will help to preserve the coefficients conjugate-symmetry when interpolating.

$$\tilde{X}(z) = \sum_{m = -\frac{N}{2}}^{\frac{N}{2}} x_m z^m \mod z^N - 1$$
(2.12)

where

$$x_{m} = \begin{cases} x[n = m] & m \in \{0, ..., N/2 - 1\} \\ x[n = m + N] & m \in \{-N/2 + 1, ..., -1\} \\ \frac{x[n = N/2]}{2} & m \in \{-N/2, N/2\} \end{cases}$$
(2.13)

An illustration of the coefficient symmetry for N=4 is presented in Figure 2.1.

¹We refer to entropy as a scalar measure of a distribution function. Specifically here, we think of all sequences as i.i.d. and uniformly distributed. For example, the entropy of a complex number is equivalent to the entropy of a pair of Real numbers.

²A slightly nicer symmetry can be reached by shifting the transform by $\sqrt{\omega}$. Specifically, this involves calculating x[n+0.5] for $n \in [N]$, and is treated in the next section.

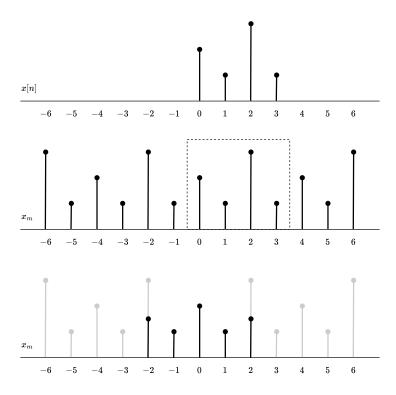


Figure 2.1: The RNTT as a Periodic Polynomial

2.4.3 Convolution

The product of two RNTT polynomials can be achieved by standard element-wise multiplication in the evaluations domain. See section 1.3.1 for more detail.

2.4.4 Interpolation

Interpolation is done in an identical manner to the standard method with a symmetry preserving zero-padding. We start by performing a length-N IRNTT over the input sequence of evaluations, outputting coefficients $\{x[n]\}_{n=0}^{N-1}$. The coefficients sequence is symmetrically zero-padded, resulting in

$$\{y[n]\}_{n=0}^{LN-1} = \{x[0], ..., x[N/2-1], \frac{x[N/2]}{2}, 0, ..., 0, \frac{x[N/2]}{2}, x[N/2+1], ..., x[N-1]\}$$

$$(2.14)$$

where the number of padded zeros depends on the interpolation factor L in the standard way. An RNTT of appropriate size is performed over the zero-padded coefficients sequence, providing the interpolated evaluations sequence. Interpolation by evaluation over individual cosets can be achieved much in the same way as presented in section 1.3.2.

An illustration of the symmetric zero-padding for N=4 and L=2 is presented in Figure 2.2.

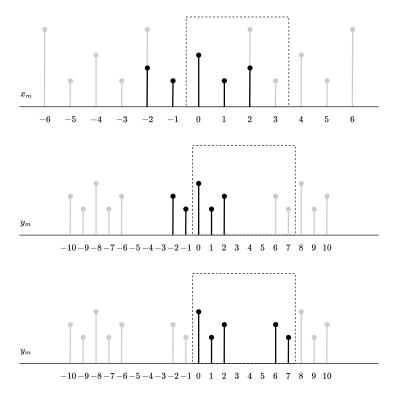


Figure 2.2: Interpolation for N=4 and L=2

2.4.5 Fast RNTT

The fast version of a RNTT (2.9) of length N involves a single CNTT of size $M = \frac{N}{2}$ and $\mathcal{O}(M)$ additional operations. For this, the only requirement is that N is even. This section is largely based on [19].

Method 1: Two Independent Sequences

The fast IRNTT for two length-M IRNTTs can be achieved using a single length-M ICNTT. For this, define the M'th root of unity η and consider the ICNTT of the complex sum of two Real sequences $X_0[k]$ and $X_1[k]$.

$$x[n] = \frac{1}{N} \sum_{k=0}^{M-1} (X_0[k] + iX_1[k]) \eta^{-kn}$$
(2.15)

$$= \frac{1}{N} \sum_{k=0}^{M-1} X_0[k] \eta^{-kn} + \frac{i}{N} \sum_{k=0}^{M-1} X_1[k] \eta^{-kn}$$
 (2.16)

$$= x_0[n] + ix_1[n] (2.17)$$

Using the conjugate-symmetry property we can determine that

$$x^*[M-n] = x_0[n] - ix_1[n]$$
(2.18)

which immediately leads to the desired result

$$x_0[n] = \frac{x[n] + x^*[M - n]}{2}$$

$$x_1[n] = \frac{x[n] - x^*[M - n]}{2i}$$
(2.19)

Note that both $x_0[n]$ and $x_1[n]$ are length-M IRNTTs. Inverting the transform is easy and can be achieved by performing a CNTT over x[n] and extracting $X_0[k]$ and $X_1[k]$ as the respective Real and Imaginary parts.

The following is an example for a length-4 IRNTT over \mathbb{F}_{19} :

```
= 19
p
    = i
mu
ΧO
    = [ 3
           9 12 2]
    = [ 7 3 10 18]
    = [ 3 9 12 2] + i[ 7
Х
                           3 10 18]
    = [7 6 4 14] + i[0]
                           4 15 9]
x/2 = [13 \ 3 \ 2 \ 7] + i[0]
    = [710410] + i[07012]
    = [0 16 15 16] + i[0 4 0 15]
x1
```

The transforms x0 and x1 were computed directly and the reader is advised to verify that relations (2.19) hold. For simplicity, x/2 is provided as well.

Method 2: A Single Sequence

Let us extend the previous two-sequence idea for the IRNTT to a single even-length-N sequence Y[k]. For this we will set $\eta = \omega^2$, where η , ω are the M-th and N-th roots of unity respectively, reminding the reader that N=2M. We begin by splitting Y[k] to two subsequences

$$X_0[k] = Y[2k] (2.20)$$

$$X_1[k] = Y[2k+1] \tag{2.21}$$

and constructing their IRNTTs $x_0[n]$ and $x_1[n]$ as before. The coefficients of y[n] can be calculated by applying the final stage of a length-N DIT IFFT as³

$$y[n] = x_0[n] + \omega^{-n} x_1[n] \tag{2.22}$$

which can be performed using the standard CT butterfly optimization. Note that $Y[k] \equiv Y(w^k)$ while $X_i[k] \equiv X_i(\eta^k) = X_i(\omega^{2k})$. Plugging (2.19) into (2.22) potentially leads to further optimization by noting that

$$y[n] = \frac{x[n] + x^*[M-n]}{2} + \omega^{-n} \frac{x[n] - x^*[M-n]}{2i}$$
 (2.23)

$$= \frac{1 - iw^{-n}}{2}x[n] + \frac{1 + iw^{-n}}{2}x^*[M - n]$$
 (2.24)

³This conclusion is easily reached by plugging the above sequence splitting into equation (2.9).

To invert the transform we can extract $x_0[n]$ and $x_1[n]$ from y[n] using the first stage of a DIF FFT as⁴

$$x_0[n] = \frac{y[n] + y[M+n]}{2}$$
 (2.25)

$$x_1[n] = \frac{y[n] - y[M+n]}{2\omega^{-n}}$$
 (2.26)

and continue as we did with the two independent sequences. Alternatively, this can be achieved directly as (TBD-fix this)

$$x[n] = x_0[n] + ix_1[n] (2.27)$$

$$= \frac{y[n] + y[M+n]}{2} + i \frac{y[n] - y[M+n]}{2\omega^{-n}}$$

$$= \frac{1 + iw^{n}}{2} y[n] + \frac{1 - iw^{n}}{2} y[M+n]$$
(2.28)

$$= \frac{1+iw^n}{2}y[n] + \frac{1-iw^n}{2}y[M+n]$$
 (2.29)

Performing an RNTT on x[n] and interleaving the Real and Imaginary parts of the result provides Y[k].

2.5Symmetric Real-valued CNTT (SNTT)

Let us define the Symmetric CNTT (SNTT) as a transform whose evaluations are Realvalued and whose coefficients obey the symmetry $x^*[n] = x[N-1-n]$, where * denotes conjugation. We will guess the inverse-transform (ISNTT) to be

$$x[n] = \frac{1}{N} \sum_{k=0}^{N-1} \tilde{X}[k] \mu^{-k(2n+1)}$$
(2.30)

$$= \frac{1}{N} \sum_{k=0}^{N-1} \mu^{-k} \tilde{X}[k] \omega^{-kn}$$
 (2.31)

where $\tilde{X}[k]$ is the SNTT of x[n] and μ is a 2N-th root-of-unity, and proceed to show that it obeys the requirements. Evidently, using a 2N-th root-of-unity for an order N transform reduces its maximal length by a factor of 2.

The symmetry is simply exhibited by

$$x^*[n] = \frac{1}{N} \sum_{k=0}^{N-1} \tilde{X}[k] \mu^{k(2n+1)} = \frac{1}{N} \sum_{k=0}^{N-1} \tilde{X}[k] \mu^{-k(2(N-1-n)+1)} = x[N-1-n]$$
 (2.32)

where we remember that $\mu^{2N} = 1$.

Let us denote by \tilde{F} the invertible matrix, whose normalized inverse is the left-multiplication inverse transform matrix then

$$\frac{1}{N}\tilde{F}^{-1} = \frac{1}{N}F^{\dagger}\Lambda^{*}$$

$$(2.33)$$

where F is the length-N CNTT matrix employing powers of ω with \dagger denoting its conjugate-transpose, and $\Lambda = \text{diag}(1, \mu, \mu^2, ..., \mu^{N-1})$. Clearly $\frac{1}{N}\tilde{F}^{-1} = \frac{1}{N}\tilde{F}^{\dagger}$, meaning that SNTT is a Fourier Transform and the forward transform can be deduced from the above as

$$\tilde{F} = \Lambda F \tag{2.34}$$

or in its equation form

$$\tilde{X}[k] = \sum_{n=0}^{N-1} x[n] \mu^{k(2n+1)}$$
(2.35)

$$= \mu^k \sum_{n=0}^{N-1} x[n] \omega^{kn}$$
 (2.36)

2.5.1 The SNTT Polynomial

If we define (2.35) as the (2N-1)-order polynomial

$$\tilde{X}(z) = \sum_{m=0}^{2N-1} x_m z^m \mod z^{2N} - 1 \tag{2.37}$$

such that $\tilde{X}[k] = \tilde{X}(z = \mu^k)$ we get

$$x_m = \begin{cases} x[n = \frac{m-1}{2}] & m \text{ is odd} \\ 0 & m \text{ is even} \end{cases}$$
 (2.38)

for $n \in [N]$. This is exactly the (2N-1)-order polynomial whose even coefficients are all zero. As such, it is completely defined by N distinct evaluations.

In what follows we will prefer to present (2.37) centered around m = 0. This will help to preserve the coefficients conjugate-symmetry when interpolating.

$$\tilde{X}(z) = \sum_{m=-N+1}^{N-1} x_m z^m \mod z^{2N} - 1$$
(2.39)

where

$$x_m = \begin{cases} x[n = \frac{m-1}{2}] & m > 0 & \text{is odd} \\ x[n = \frac{m-1}{2} + N] & m < 0 & \text{is odd} \\ 0 & m \text{ is even} \end{cases}$$
 (2.40)

2.5.2 Convolution

The product of two SNTT polynomials can be achieved by standard element-wise multiplication in the evaluations domain. See section 1.3.1 for more detail.

2.5.3 Interpolation

Interpolation is done in an identical manner to the standard method with a symmetry preserving zero-padding. We start by performing a length-N ISNTT over the input sequence of evaluations, outputting coefficients $\{x[n]\}_{n=0}^{N-1}$. The coefficients sequence is symmetrically zero-padded, resulting in

$$\{x[0], ..., x[N/2-1], 0, ..., 0, x[N/2], ..., x[N-1]\}$$
 (2.41)

where the number of padded zeros depends on the interpolation factor in the standard way. An SNTT of appropriate size is performed over the zero-padded coefficients sequence, providing the interpolated evaluations sequence. Interpolation by evaluation over individual cosets can be achieved much in the same way as presented in section 1.3.2.

2.5.4 Fast SNTT

Fast SNTT and ISNTT can be performed using Fast RNTT methods. To do this we use equations (2.34) and (2.33) to construct the following SNTT and ISNTT relations

$$\mathbf{X}_0 + i\mathbf{X}_1 = \Lambda F(\mathbf{x}_0 + i\mathbf{x}_1) \tag{2.42}$$

$$\mathbf{x}_0 + i\mathbf{x}_1 = \frac{1}{N} F^{\dagger} \Lambda^* (\mathbf{X}_0 + i\mathbf{X}_1)$$
 (2.43)

respectively. This immediately enables using the methods from Section 2.4.5.

Chapter 3

Additive NTT

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3.1 Introduction

Extension fields of characteristic two are of vast importance. One paramount use-case is efficient encoding and decoding of low-rate Reed-Solomon codes over those fields [20]. Standard methods relay on the Fast Fourier Transform (FFT) algorithm that achieves $\mathcal{O}(n \log n)$ complexity. The problem is that FFTs employ a multiplicative subgroup of the field and are typically most efficient for subgroups whose size is a power-of-two. For binary extension fields, the multiplicative group $\mathbb{F}_{2^n}^*$ is of odd size and thus has no roots of unity whose order is a power-of-two.

Additive FFTs over finite extension fields started showing up in the late 1980's [21]. An additive FFT, which we will term Additive NTT (ANTT), is an evaluation over an additive subgroup, rather than over a multiplicative subgroup. Interestingly, they are not Fourier Transforms at all, but they obey the recursive FFT-like construction, achieving $\mathcal{O}(n \log n)$ complexity.

This chapter aims to provide a simplified, tutorial-styled exposition on the subject. We start by presenting the construction and then dive into the mathematical basis for it. We present the construction for the general case of characteristic two and provide explicit formulations for two important cases.

3.2 NTT over an Additive Subgroup

Consider a representation of the field elements of \mathbb{F}_{2^n} as

$$\omega_i = \sum_{j=0}^{n-1} i_j \nu_j \tag{3.1}$$

where i_j is the j'th bit of i, and $\{\nu_j\}_{j=0}^{n-1}$ is some vector space \mathbb{F}_2^n . We note that $\omega_0 = 0$ and that $\{\omega_i\}_{i=0}^{2^k-1}$, $k \leq n$ is an additive subgroup of \mathbb{F}_{2^n} . The ANTT of size 2^k is defined as the evaluation of a polynomial $P(x) \in \mathbb{F}_{2^n}[x]/x^{2^k} - x$ over an additive subgroup coset $\{\omega_i + \alpha\}_{i=0}^{2^k-1}$ where $\alpha \in \mathbb{F}_{2^n}$. Note that $\{\omega_i\}_{i=0}^{2^k-1}$ are exactly the 2^k roots of $x^{2^k} - x$. Since

 $\{\omega_i + \alpha\}_{i=0}^{2^k-1}$ is a coset of the additive subgroup $\{\omega_i\}_{i=0}^{2^k-1}$ and is in \mathbb{F}_{2^n} , we can construct a new polynomial $\widetilde{P}(x) \in \mathbb{F}_{2^n}[x]/x^{2^n} - x$ by zero-padding P(x) and then evaluate it over \mathbb{F}_{2^n} . The ANTT over the coset is a subset of this.

To facilitate a fast ANTT we limit our scope to binary tower fields. Binary tower fields [22] are a nested field series such that the first field in the series is \mathbb{F}_2 and each subsequent field is a quadratic extension of its predecessor. This necessarily means that n, the size of its basis, is a power-of-two. The restriction to binary tower fields is required for the construction of the *subspace* polynomials as discussed in Section 3.4 hereafter.

3.3 Fast ANTT

To enable a fast ANTT, let us first define a configuration of P(x) in the vector space spanned by $\{X_{\ell}(x)\}_{\ell=0}^{2^{n}-1}$, the *novel* polynomial basis [23],

$$P(x) = \sum_{\ell=0}^{2^{k}-1} p_{\ell} X_{\ell}(x)$$
 (3.2)

where we emphasize that $deg(X_{\ell}(x)) = \ell$. Using this basis, we show how $P(\omega_i + \alpha)$ can be computed recursively, and then use this recursion to construct the fast ANTT algorithm. Before presenting the recursion, let us define the *novel* basis vectors as

$$X_{\ell}(x) = \prod_{j=0}^{n-1} \left(\frac{s_j(x)}{s_j(\omega_{2^j})} \right)^{\ell_j}$$
 (3.3)

where ℓ_j is the j'th bit of ℓ . Note that the term inside the product is normalized so that its value at ω^{2^j} is 1. The *subspace* polynomials $\{s_j(x)\}_{j=0}^{n-1}$ are defined as the subgroup vanishing polynomials

$$s_j(x) = \prod_{i=0}^{2^j - 1} x + \omega_i \tag{3.4}$$

that obey the following polynomial linearity

$$s_i(x+y) = s_i(x) + s_i(y)$$
 (3.5)

The exact construction is deferred to Section 3.4.

The recursion for an ANTT of size 2^k (i.e. $i \in [2^k]$) proceeds as follows (Define $\tau \equiv 2^{k-1}$ for notational ease)

$$P(\omega_i + \alpha) = \sum_{\ell=0}^{2^k - 1} p_\ell X_\ell(\omega_i + \alpha)$$
(3.6)

$$= \sum_{\ell=0}^{\tau-1} p_{\ell} X_{\ell}(\omega_i + \alpha) + \sum_{\ell=0}^{\tau-1} p_{\tau+\ell} X_{\tau+\ell}(\omega_i + \alpha)$$
 (3.7)

$$= \sum_{\ell=0}^{\tau-1} p_{\ell} X_{\ell}(\omega_i + \alpha) + \sum_{\ell=0}^{\tau-1} p_{\tau+\ell} \frac{s_{k-1}(\omega_i + \alpha)}{s_{k-1}(\omega_{\tau})} X_{\ell}(\omega_i + \alpha)$$
 (3.8)

3.3. FAST ANTT 25

$$= \sum_{\ell=0}^{\tau-1} (p_{\ell} + \frac{s_{k-1}(\omega_i + \alpha)}{s_{k-1}(\omega_{\tau})} p_{\tau+\ell}) X_{\ell}(\omega_i + \alpha)$$
(3.9)

$$= \begin{cases} \sum_{\ell=0}^{\tau-1} (p_{\ell} + \frac{s_{k-1}(\alpha)}{s_{k-1}(\omega_{\tau})} p_{\tau+\ell}) X_{\ell}(\omega_{i} + \alpha) & i < \tau \\ \sum_{\ell=0}^{\tau-1} (p_{\ell} + \frac{s_{k-1}(\alpha)}{s_{k-1}(\omega_{\tau})} p_{\tau+\ell} + p_{\tau+\ell}) X_{\ell}(\omega_{i-\tau} + \omega_{\tau} + \alpha) & i \geq \tau \end{cases}$$
(3.10)

where (3.8) follows from (3.3) since $X_{\tau+\ell}(x) = \frac{s_{k-1}(x)}{s_{k-1}(\omega_{\tau})} X_{\ell}(x)$, and (3.10) can follows by

$$\omega_i = I(i \geqslant \tau)\omega_\tau + \omega_{(i \mod \tau)} \tag{3.11}$$

$$= \begin{cases} \omega_i & i < \tau \\ \omega_{i-\tau} + \omega_{\tau} & i \geqslant \tau \end{cases}$$
 (3.12)

and

$$s_{k-1}(\omega_i + \alpha) = s_{k-1}(\omega_i) + s_{k-1}(\alpha)$$
(3.13)

$$= \begin{cases} s_{k-1}(\alpha) & i < \tau \\ s_{k-1}(\alpha) + s_{k-1}(\omega_{\tau}) & i \geqslant \tau \end{cases}$$

$$(3.14)$$

The recursion leads to fast algorithms for ANTT and Inverse ANTT (IANTT) as presented in Algorithms 1 and 2, respectively.

```
Algorithm 1 ANTT(P(x), \alpha, k)
```

```
1: if k=0 then
  2:
            return p_0
  3: end if
  5: \tau \leftarrow 2^{k-1}
 6: P_0(x) \leftarrow \sum_{\ell=0}^{\tau-1} p_{\ell} X_{\ell}(x)

7: P_1(x) \leftarrow \sum_{\ell=0}^{\tau-1} p_{\tau+\ell} X_{\ell}(x)

8: Q_0(x) \leftarrow P_0(x) + \frac{s_{k-1}(\alpha)}{s_{k-1}(\omega_{\tau})} P_1(x)

9: Q_1(x) \leftarrow Q_0(x) + P_1(x)
11: return concatenate(ANTT(Q_0(x), \alpha, k-1), ANTT(Q_1(x), \omega_\tau + \alpha, k-1))
```

Algorithm 2 IANTT $(P(\{\omega_i\}_{i=0}^{2^k-1} + \alpha), \alpha, k)$

```
Input: [P(\omega_{0} + \alpha), P(\omega_{1} + \alpha), ..., P(\omega_{2^{k}-1} + \alpha)] \in \mathbb{F}_{2^{n}}^{2^{k}}, \alpha \in \mathbb{F}_{2^{n}}, k \leq n

Output: P(x) = \sum_{\ell=0}^{2^{k}-1} p_{\ell} X_{\ell}(x), [p_{0}, p_{1}, ..., p_{2^{k}-1}] \in \mathbb{F}_{2^{n}}^{2^{k}}

1: if k=0 then

2: return P(\omega_{0} + \alpha)

3: end if

4:

5: \tau \leftarrow 2^{k-1}

6: Q_{0}(x) \leftarrow \text{IANTT}(P(\{\omega_{i}\}_{i=0}^{\tau-1}, \alpha, k-1)

7: Q_{1}(x) \leftarrow \text{IANTT}(P(\{\omega_{\tau+i}\}_{i=0}^{\tau-1}, \omega_{\tau} + \alpha, k-1)

8: P_{1}(x) \leftarrow Q_{0}(x) + Q_{1}(x)

9: P_{0}(x) \leftarrow Q_{0}(x) + \frac{s_{k-1}(\alpha)}{s_{k-1}(\omega_{\tau})} P_{1}(x)

10:

11: return P_{0}(x) + \frac{s_{k-1}(x)}{s_{k-1}(\omega_{\tau})} P_{1}(x)

\Rightarrow concatenation of polynomial coefficients
```

3.4 Subspace Polynomials and the Field Basis

By equation (3.4), the two first *subspace* polynomials are uniquely determined for any basis of \mathbb{F}_{2^n}

$$s_0(x) = x \tag{3.15}$$

$$s_1(x) = x^2 + x (3.16)$$

by definition of the trivial subgroup $\{\omega_0\}$ and the binary field \mathbb{F}_2 . Furthermore, integrating the polynomial linearity (3.5) into equation (3.4) leads to the following recursion

$$s_{j+1}(x) = \prod_{i=0}^{2^{j+1}-1} x + \omega_i$$
(3.17)

$$= \prod_{i=0}^{2^{j}-1} (x + \omega_i) \prod_{i=0}^{2^{j}-1} (x + \omega_i + \omega_{2^{j}})$$
 (3.18)

$$=s_j(x)s_j(x+\omega_{2^j})\tag{3.19}$$

$$= s_j^2(x) + s_j(\omega_{2^j})s_j(x)$$
 (3.20)

which can be presented using the following geometric composition

$$\phi_j(x) = x^2 + \beta_j x \tag{3.21}$$

where $\beta_j \equiv s_j(\omega_{2^j})$.

Finally, we can present the *subspace* polynomial series compactly as

$$s_0(x) = x$$

 $s_j(x) = \phi_j(\phi_{j-1}(....\phi_1(x)))$ (3.22)

Since $s_0(x) = x$ and due to the structure of the composition (3.21), it's easy to show that all *subspace* polynomials are of the form

$$s_j(x) = \sum_{i=0}^{j} a_{j,i} x^{2^i}$$
(3.23)

Polynomial linearity (3.5) can be verified using the Frobenius endomorphism [24] (a.k.a. Freshman's Dream) on equation (3.23) since for characteristic two, $(x + y)^2 = x^2 + y^2$.

Correctness of (3.22) everywhere on \mathbb{F}_{2^n} entails correctness for its basis vectors $\{\nu_j\}_{j=0}^{n-1}$. Specifically if we set $\nu_0 = 1$ then the basis vectors must adhere to a recursive construction obeying the *subspace* polynomials structure. We now present two such constructions.

3.4.1 The Cantor Basis

The Cantor basis [25] can be constructed by setting

$$\beta_j = s_j(\omega_{2^j}) = s_j(\nu_j) = 1$$
 (3.24)

which leads to the geometric composition functional

$$\phi_j(x) = \phi(x) = x^2 + x \tag{3.25}$$

resulting in the *subspace* polynomials

$$s_j(x) = \begin{cases} x & j = 0\\ \phi^{\circ j}(x) & j > 0 \end{cases}$$
 (3.26)

where $\phi^{\circ j}$ denotes the composition of ϕ , j times. Recursively, this means that

$$s_{j+1}(x) = s_j^2(x) + s_j(x)$$
(3.27)

By induction on (3.27) it is easy to show that

$$s_{j}(x) = \sum_{m=0}^{j} {j \choose m} x^{2^{m}}$$
(3.28)

using the binomial property

and noting that the binomial coefficients in this case are binary (i.e. $\binom{j}{m} \in \mathbb{F}_2 \Rightarrow \binom{j}{m}^2 = \binom{j}{m}$).

Equation (3.24) can be represented as

$$\phi^{\circ j}(\nu_j) = \nu_0 \tag{3.30}$$

using (3.26) and the fact that $\nu_0 = 1$. Choosing the basis elements recursively as

$$\nu_j^2 + \nu_j = \nu_{j-1} \tag{3.31}$$

results in the Cantor tower field construction as is shown in the Appendix of [26].

3.4.2 The Binius Basis

The Binius basis [27] is constructed by the following recursion

$$\nu_0 = 1 \tag{3.32}$$

$$\nu_{2^{\kappa}} + \nu_{2^{\kappa}}^{-1} = \nu_{2^{\kappa} - 1} \tag{3.33}$$

Using (3.22) it is possible to construct an ANTT for this basis. Experimentally, it was found that this basis adheres to the following β series (indexing starts from 0)

$$[1, 1, 3, 2, 9, 6, 5, 8, 65, 168, 114, 40, 175, 152, 51, \dots]$$

where $\beta_j = s_j(\omega_{2^j})$. Investigating the structure of the resulting series and is left to the interested reader.

3.5 Is ANTT a Fourier Transform?

We can easily show that ANTT is not a Fourier Transform by examining the matrix form of the transform for size 4. In this case, it is easy to verify that both Cantor and Binius bases are identical. The ANTT matrix is simply the following evaluation

$$F = \begin{bmatrix} X_0(0) & X_1(0) & X_2(0) & X_3(0) \\ X_0(1) & X_1(1) & X_2(1) & X_3(1) \\ X_0(2) & X_1(2) & X_2(2) & X_3(2) \\ X_0(3) & X_1(3) & X_2(3) & X_3(3) \end{bmatrix} = \begin{bmatrix} 1 & 0 & 0 & 0 \\ 1 & 1 & 0 & 0 \\ 1 & 2 & 1 & 2 \\ 1 & 3 & 1 & 3 \end{bmatrix}$$
(3.34)

In the Fourier Transform case, the IANTT matrix is the transpose of F. If we multiply F by its transpose F^T , we get

$$F \cdot F^{T} = \begin{bmatrix} 1 & 1 & 1 & 1 \\ 1 & 0 & 3 & 2 \\ 1 & 3 & 0 & 0 \\ 1 & 2 & 0 & 0 \end{bmatrix} \neq I$$
 (3.35)

where I is the identity matrix. But since $F \cdot F^T \neq I$ clearly F^T is not the IANTT matrix and hence ANTT is not a Fourier transform.

Chapter 4

Implementation on FPGA

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4.1 Introduction

NTT implementation is cumbersome. The main challenge is data movement between memories and processors. For small NTTs, whose data completely fits in on-chip random access memory, it is usually easy to build efficient solutions. The complications arise when NTT sizes are too large and data must be handled via off-chip memory such as DDR or HBM [28, 29]. To illustrate this we chose to implement a hardware accelerator for the maximumsize NTT used in the Filecoin protocol [30]. This NTT is of length 2²⁷ over elements of the scalar field of BLS12-381 [31] with an appropriate root-of-unity. An element of the scalar field is 255 bits long, but we will use 256 bits or 32 Bytes for our calculations hereafter. The platform we chose to use for this implementation is Xilinx's C1100 card which houses the Ultrascale+ FPGA device VU35P [32, 33]. We chose this platform since it is readily available and houses an HBM device. As it turns out, most of the challenge is in matching the NTT's parameters to the platform's restrictions such that the trade-offs between interface, memory, and processing are optimized. Although our design choices are fairly customized to the selected operating point, we paid careful attention to architect the solution in an extendable manner allowing future migration to other operating points. The main purpose of this chapter is to describe in fair detail the architecture and design process for building a large NTT efficiently. Although this chapter was written after the fact, it attempts to describe the architecture and design process as a chronological decision-making process similar to the one that took place in practice.

4.2 Solution Landscape

Our hardware accelerator resides in a CPU-PCIe system. This means that the C1100 card is installed on a PCIe bus as a companion card in a CPU environment. NTT tasks are initiated by the CPU by streaming data from the PC memory to the FPGA. Once

completed, the output is streamed back from the FPGA to the PC memory. The complete sequence can be broken down into the following stages:

- 1. PC streams input data to HBM
- 2. HBM data is read to FPGA fabric
- 3. Data is processed in FPGA fabric
- 4. Processed data is written back to HBM
- 5. PC streams output data from HBM

Note that steps 2 to 4 may repeat multiple times.

We can break down the above list into three distinct mechanisms that can run concurrently. Lines 1 and 5 involve PCIe transactions for moving data in and out of the accelerator. Lines 2 and 4 involve HBM transactions for moving data in and out of the processing core. Finally, line 3 involves the processing itself. The following three subsections describe limiting factors for the three mechanisms that greatly influenced our design choices, followed by a fourth subsection that provides our initial throughput analysis, given these choices.

4.2.1 PCIe to HBM

A single NTT's input size in Bytes is $2^{27} \cdot 32 = 2^{32}$. That is 4GB which is exactly half of the 8GB HBM capacity in VU35P. Assuming in-place NTT processing, this allows storing the data for two complete NTTs concurrently. The first design choice we made was to use the HBM as a double buffer. This allows processing the current NTT in one buffer while using the other buffer to stream the previous NTT out and the next NTT in. The double buffer structure allows further extensions discussed later in the chapter. Figure 4.1 illustrates the double buffer mechanism.

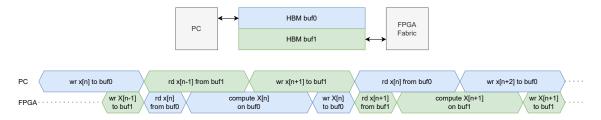


Figure 4.1: HBM Double Buffer

The choice of PCIe type (generation and number of lanes) was made to provide reasonable throughput while keeping the PCIe controller sufficiently small. This is important since for VU35P the PCIe controller is instantiated on FPGA fabric and consumes resources that can otherwise be used for processing. Our initial choice was PCIe Gen3x8 with optional future upgrades.

4.2.2 HBM to Fabric

In order to prevent the HBM interface from becoming a bottleneck, it is important to structure it to provide sufficient parallel throughput. The main trade-off is again the area occupancy of the HBM control logic in FPGA fabric. The HBM in VU35P is subdivided into 8 banks of 1GB each. Each bank is attached to a 4×4 switch. Each switch connects to its 1GB HBM bank with four 64 bits, independently addressable, full-duplex interfaces, each running at 1.8GSps. Each of the four outputs of each switch can be connected to an AXI master with configurable width. In order to match our requirements, we configured the switch's outputs to be 256 bits wide. This lowered the interface speed to 450MSps, maintaining the throughput. As we shall see, The logic required to translate the interface from 64 to 256 bits limited the total number of 256-bit AXI interfaces per switch to two, reducing the total possible throughput to half of its maximum capacity. As we shall see, this is still sufficiently high to not become a bottleneck. An additional mechanism offered in the VU35P device is an interconnecting network between the switches. This allows any AXI master to address any of the eight HBM banks, regardless of the switch it is connected to. Since the NTT access pattern is quite regular, we decided to disable this logic and only allow each master to access the 1GB bank directly attached to its switch. In fact, we further partitioned the 1GB banks into two 0.5GB banks such that each AXI master has its own distinct address space.

The resulting HBM interface is 16 AXI masters, each 256-bit wide at a maximum speed of 450MSps. going forward let us use *Word* to describe a 256-bit data word. For efficiency, we would like to read and write the HBM in whole pages only. The HBM page size is 1KB which is 32 Words, so each AXI master's 0.5GB is an address space consisting of 2¹⁹ pages, which is 2¹⁸ pages per buffer in the double-buffer scheme described before. Each AXI master is equipped with a Memory Management Unit (MMU) that is used to control the internal data-flow as described later in the chapter. Figure 4.2 illustrates the resulting HBM interface.

Notably, the connection of the PCIe controller to the HBM is also routed via the switches. Focusing mainly on the internal interaction, our initial design elected the simplest solution of connecting the PCIe via a single AXI master interface connected to the physically closest switch. For this master, we allowed the cross-connecting feature between switches such that it can access the whole HBM space. This design choice is obviously limiting as it provides limited bandwidth for the PCIe and potentially causes congestion due to the interconnect activity. It is our intention to improve this in the future along with additional extensions for the PCIe MMU as is described later in this chapter.

4.2.3 Processing in Fabric

The core data processor is perhaps the main topic for much of this chapter. Since it significantly affects the rest of the design, its early (and accurate) selection is crucial.

Understanding that most of the processing involves 256-bit modular multiplications, we opted to estimate how many such multipliers we would be able to instantiate without pushing the design beyond the physical boundaries of the device. We chose to use our single precision Domb-Barrett multipliers [34] and estimated that for this particular field, we would be able to fit approximately 12 such multipliers. A less connected design could perhaps fit a little more but we estimated that the required data-flow, and particularly the

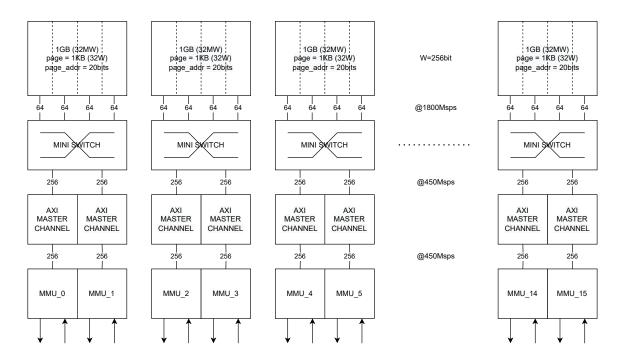


Figure 4.2: HBM Interface

calculation of what we will soon term subNTT, will restrict us to fit the main processor logic within a single FPGA SLR^1 leading to the more conservative limit.

Another aspect of selecting the processor core involves the number of round trips required for the NTT data from and to the HBM. Using the Cooley-Tukey (CT) algorithm for Fast Fourier Transform (FFT) leads to performing a number of stages (or cycles) of processing over the whole NTT data. The processing per stage is very similar across stages and involves passing the data through a processor called a *Butterfly* in equally sized chunks that completely partition the data. The size of the chunk is usually called the radix of the butterfly. An FFT can be performed such that all stages use the same radix or by utilizing different radixes in which case it is termed mixed-radix. The number of stages is directly determined by the radix (or radixes). For a single-radix design, the number of stages is the logarithm of the NTT size with the radix used as the base. As such, a larger radix results in fewer stages, thus fewer round trips from and to the HBM. Another advantage of a larger radix is the ability to use the Winograd FFT algorithm, potentially saving multipliers [8].

The downside of a large radix is that it complicates the element ordering and transposition requirements for FFT processing much more than for the ordinary radix-2, as we shall see later in this chapter.

After careful consideration, we opted to work with a radix-8 core. As we shall see, the cost of this core in the Winograd implementation is eight full multipliers and four constant multipliers. Our analysis shows that it should fit into a single SLR and enable constructing our subNTT on top of it without over-complication. Other alternatives that were considered and discarded were parallel radix-2 engines and parallel radix-4 engines. A single radix-16 engine was very suitable but unfortunately not feasible due to its size.

 $^{^{1}\}mathrm{VU}35\mathrm{P}$ has two SLRs and SLR crossings across wide data-flow interfaces greatly limits the maximum working frequency.

The smaller radix options proved to not match well with the 16 HBM MMU design or over-complicated the transposition logic.

The simple way to work with 16 HBM MMUs and radix-8 cores is to split the data between the left 8 and right 8 MMUs and have a radix-8 core per side. Since only a single radix-8 core is available we decided to double-clock it to serve both HBM sides.

4.2.4 Throughput Estimation

The table in Figure 4.3 shows the throughput analysis for different versions of the NTT accelerator. The difference between the versions is the operating frequencies of various parts in the design. Note how the PCIe interface and fabric processing dominate the total time, while the HBM is less significant. Although this was not a prior consideration, this is very desirable from a power dissipation perspective.

		V1	V2	V3	V4	V5
number of nttcs		1	1	1	1	1
nttc radix		8	8	8	8	8
Word size	Bytes	32	32	32	32	32
ntt size	Words	134217728	134217728	134217728	134217728	134217728
subntt size	Words	512	512	512	512	512
subntts per stage		262144	262144	262144	262144	262144
number of stages		3	3	3	3	3
hbm axi freq	MHz	125	125	250	250	250
hbm freq (net)	MHz	93.75	93.75	187.5	187.5	187.5
ntt load/store size	Words	402653184	402653184	402653184	402653184	402653184
hbm thruput	Words/tx	16	16	16	16	16
total tx/rx time (full duplex)	s	0.268	0.268	0.134	0.134	0.134
nttc freq	MHz	125	250	500	500	500
radix8 substages per subntt		3	3	3	3	3
extra radix8 substage for tf		3	3	3	3	3
total radix8 substages for all stages		12	12	12	12	12
total radix8 clk for one subntt for all stages		768	768	768	768	768
total radix8 clk for one ntt		201326592	201326592	201326592	201326592	201326592
total radix8 clk for ntt across nttcs		201326592	201326592	201326592	201326592	201326592
total time per ntt s		1.61	0.81	0.40	0.40	0.40
		gen3x8	gen3x8	gen3x8	gen4x8	gen4x16
pcie thruput (full duplex)	GB/s	8	8	8	16	32
pcie thruput (net)	GB/s	6	6	6	12	24
ntt size in Bytes	Bytes	4294967296	4294967296	4294967296	4294967296	4294967296
total time per ntt	s	0.72	0.72	0.72	0.36	0.18

Figure 4.3: Throughput Estimation

4.3 Solution Scheme

One way to illustrate the single-radix CT scheme is by placing the NTT data on a hypercube². The dimension and edge-length of the hypercube are the number of required stages and the radix, respectively. For the simple case of two stages, the reader is referred back to Section 1.4. In this case, we decided to split the 2^{27} into three stages of radix-512, where each 512 NTT is termed subNTT. The scheme follows the illustration in Figure 4.4.

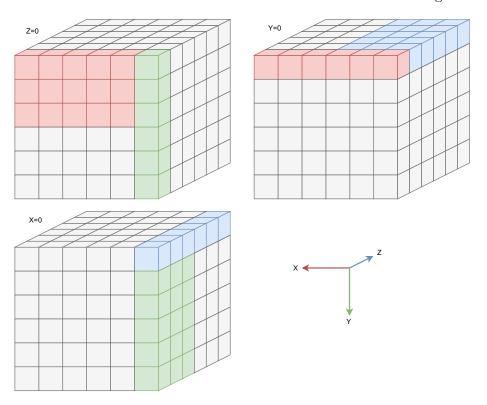


Figure 4.4: CT FFT Cube Scheme

We initially organize the data over a three-dimensional cube. The data organization can be done in one of two arrangements (more on this soon). At the first stage, we perform all subNTTs along the x-axis, then at the second stage along the y-axis, and finally at the third stage along the z-axis. Between stages, we multiply by the appropriate twiddle-factors. Note that from a data-flow perspective, switching between two axes, as is done between stages, is equivalent to a transpose operation between those two axes.

Figure 4.5 is a visualization of the same scheme from a slightly different perspective. In this visualization, we tried to number the samples in the same way as they appear in memory for the in-place implementation. The ordering is quite cumbersome but one important thing to notice is how the order of subNTTs at the output is strided relative to its sequential order at the input. The striding is by a factor of 512 and is actually equivalent to reversing the bit order of the 9 least significant bits in the subNTT 18-bit indexes. This is a characteristic of the in-place CT FFT. The subNTT level ordering is either strided at the input for DIF or output for DIT. Note that this is independent of the

²Or hyper-rectangle for mixed-radix schemes.

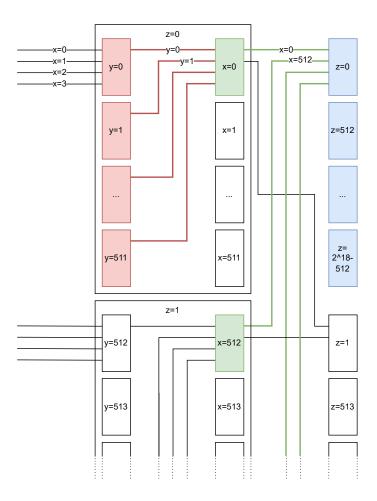


Figure 4.5: CT FFT Butterfly Scheme

ordering of inputs and outputs of the subNTT itself.

We noted, through experimentation, that the simplest way (for us) to systematically map the transposed process, without requiring bit-reverse ordering of the data at every stage was to keep the data reverse-bit-ordered throughout the whole process. The outcome of that was that we constructed every NTT level, including radix-8, subNTT, and the full NTT to be reverse-bit-ordered at the input and output. The only outcome is that it required calculating the twiddle-factors in a reverse-bit-ordered fashion as $\omega^{\text{rbo}(\rho)\text{rbo}(l)}$ rather than $\omega^{\rho l}$. In conclusion, the input to the NTT accelerator must be reverse-bit-ordered by the PC prior to streaming to the device. When received back it must be subNTT block reordered to remove the striding and then reverse-bit-ordered. Clearly, all reordering can be done in $\mathcal{O}(n)$. Moving some of the reordering to hardware is a topic that is currently being explored by us.

4.4 Architecture

We mentioned in a previous section that the number of MMUs is 16 while the internal processor is a single radix-8 machine (i.e. it has eight inputs). As mentioned there, the

way to make this work is by time-sharing the radix-8 processor between the left and right MMUs. To simplify the presentation in this section, let us ignore the time-sharing and pretend that we have two radix-8 processors.

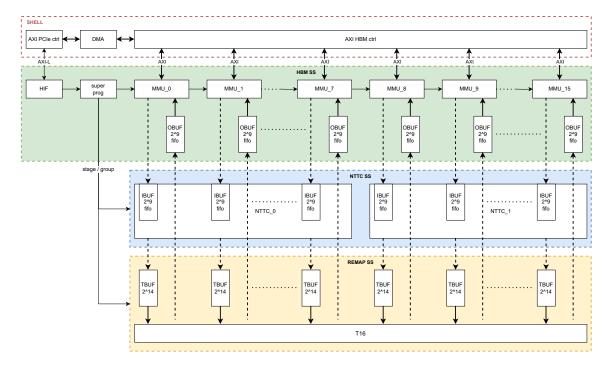


Figure 4.6: Top Level Architecture

Let us begin by presenting the top-level architecture (see Figure 4.6). The presentation uses a data grouping terminology that was not yet defined but will be clarified as the section proceeds. The design is split into four subsystems, as follows:

- SHELL Consists of the PCIe (and DMA) interface to the PC and the HBM AXI interfaces to the fabric.
- HBM-SS Consists of the HBM MMUs and the super-program state-machine.
- NTTC-SS Consists of the logic required to compute a *Batch* of subNTTs, including multiplication by external twiddle-factors required for the full NTT.
- REMAP-SS Consists of the logic required to transpose a *Slice* of subNTTs.

The design is controlled by the super-program state-machine that essentially manages the data-flow and processing across all subsystems. The general flow of the super-program, as depicted in Figure 4.7, is as follows:

- 1. Once data is available in HBM, the HBM-SS commences the super-program.
- 2. MMUs move a Batch of data from HBM to IBUFs (a.k.a. input buffers). A Batch is 16 subNTTs, sub-grouped according to even and odd indices. The eight even indexed subNTTs are at the left MMUs (i.e. 0 to 7) and the eight odd subNTTs at the right MMUs (i.e. 8 to 15). The data is written to IBUFs sequentially.

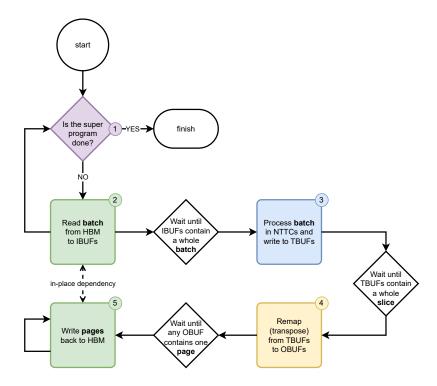


Figure 4.7: Super Program State Machine

- 3. The two NTTCs within NTTC-SS process the data Batch, reading from IBUFs and writing to TBUFs (a.k.a. transpose buffers). The data is read from IBUFs and written to TBUFs sequentially.
- 4. Once TBUFs contain at least one Slice, which is 16 Batches, T16 within the REMAP-SS transposes the data in TBUFs and writes the transposed data to OBUFs (a.k.a. output buffers). The data is written to OBUFs sequentially. The data to an OBUF has to be in a minimal quant of an HBM page (32 Words). This assures that the HBM transaction overheads are minimal. Slice is the minimum amount of data required for transposing into whole pages at the output.
- 5. MMUs write pages of data from OBUFs back to HBM.
- 6. This process repeats for all three stages and all Batches per stage. The subsystems' functional configuration changes across stages and Batches.

For all intents and purposes, a single instruction of the super-program is used to process a Group which amounts to two consecutive Slices. This minimizes internal stalls that are not compute-induced. A complete 2^{27} NTT will run a super-program going through all stages and all Groups in order, but running individual Groups in random order or any sub-programs or combinations thereof is possible. This is often helpful for debugging purposes.

In order to keep the design as modular as possible without sacrificing increased delays due to a lack of synchronization between subsystems, the design data-flow is controlled using back-pressure techniques. Once a super-program has commenced, the data will

continuously flow through the design until the program is complete. Each block will digest incoming data or stall it if it is currently busy. This assures no unnecessary stalls. The back-pressure scheme is portrayed in Figure 4.8.

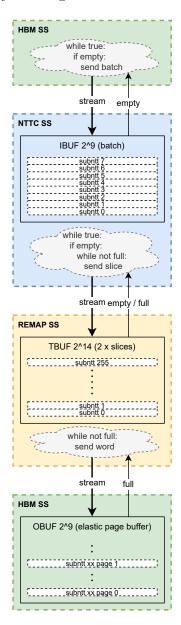


Figure 4.8: Back-Pressured Data-Flow Design

The next subsections outline the data organization and architectural details for the main functions in the design.

4.4.1 Data Organization

The data is organized in the HBM in the order in which it is read into the design. The general organization is depicted in Figure 4.9.

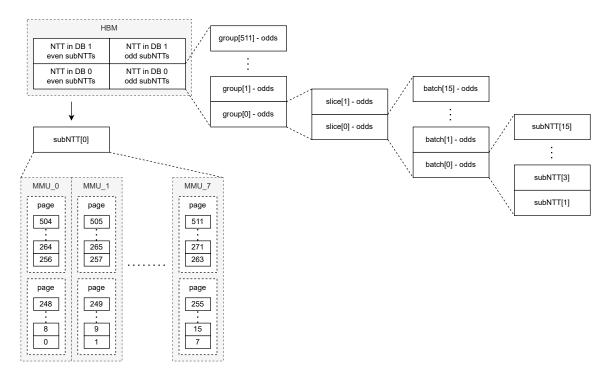


Figure 4.9: Data Organization in HBM

In essence, the HBM is subdivided into two rows and two columns. The two rows account for the HBM double buffer and the two columns account for the left and right NTTC sides. The NTT data (corresponding to a single buffer) consists of 512 Groups, each Group consisting of two Slices, each Slice consisting of 16 Batches, and each Batch consisting of 16 subNTTs. The even indexed subNTTs in a Batch are placed on the left NTTC side and the odd on the right. The lower part of Figure 4.9 shows how a single subNTT of 512 Words is distributed across the left eight MMUs. Note that this results in two pages per MMU (out of 8 MMUs of a particular NTTC side) for storage of each subNTT.

An alternative visualization of the data organization is presented in Figure 4.10 where the data is shown in a pyramid structure for a single complete NTT. The left pyramid in the figure presents the organization from the subNTT perspective while the right pyramid presents it from the page perspective.

4.4.2 MMU (HBM-SS)

The MMUs are primarily in charge of selecting the groups flowing from HBM to the processor and back to HBM. If we refer back to Figure 4.5, the MMUs are in charge of selecting the red group at stage 0, the green group at stage 1, and the blue group at stage 2. Per stage, the groups are selected according to the transpose input requirements of T16. It is clear from the figure that T16 performs the FFT wiring between stages (i.e. green wiring between stages 0 and 1 and red wiring between 1 and 2). The table in Figure 4.11 provides a more detailed view of the MMUs read/write ordering.

The table presents the order of subNTTs on and between stages for the whole super-

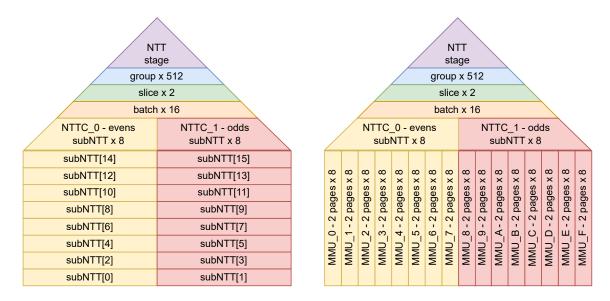


Figure 4.10: Data Organization in HBM - subNTT (left) vs. pages (right) perspectives

		I		read [0]	natural	read [1]	strided	read [2]	natural	read out	natural	1	1
				write [1]	grp flip	write [2]	strided	write [3]	grp flip	read out	Haturai		
				write [1]	gip iiip	wite [E]	ounded	Write [3]	grp mp				
block = 6	E4 Worde	grp index	slice index	[D]epcts	raw data	ete	000[1]	eta	ge[2]	output pro	cessed data	output eu	bNTT order
block = 64 Words grp index slice index block index block addr for stages 0 & 2		stage[0] - raw data NTTC 0 NTTC 1		NTTC 0			NTTC 1	NTTC 0	NTTC 1	NTTC 0	NTTC 1		
0	0	0	0	0	1	0	1	0	1	0	1	0	512
1	64	"		2	3	2	3	2	3	2	3	1024	1536
2	128			4	5	4	5	4	5	4	5	2048	2560
3						T	-				7		
4	192 256			6 8	7 9	6	7 9	6	7 9	6	9	3072 4096	3584 4608
5	320			10	11	8	11	8 10	11	8 10	11		5632
6						10 12	I	12				5120	
	384			12	13		13		13	12	13	6144	6656
7	448			14	15	14	15	14	15	14	15	7168	7680
				<u></u> .		<u></u> .							
127	8128			254	255	254	255	254	255	254	255	130048	130560
128	8192		1	256	257	256	257	256	257	256	257	131072	131584
255	16320			510	511	510	511	510	511	510	511	261120	261632
256	16384	1	2	512	513	513	512	513	512	512	513	1	513
257	16448			514	515	515	514	515	514	514	515	1025	1537
258	16512			516	517	517	516	517	516	516	517	2049	2561
259	16576			518	519	519	518	519	518	518	519	3073	3585
260	16640			520	521	521	520	521	520	520	521	4097	4609
261	16704			522	523	523	522	523	522	522	523	5121	5633
262	16768			524	525	525	524	525	524	524	525	6145	6657
263	16832			526	527	527	526	527	526	526	527	7169	7681
383	24512			766	767	767	766	767	766	766	767	130049	130561
384	24576		3	768	769	769	768	769	768	768	769	131073	131585
511	32704			1022	1023	1023	1022	1023	1022	1022	1023	261121	261633
512	32768	2	4	1024	1025	1024	1025	1024	1025	1024	1025	2	514
639	40896			1278	1279	1278	1279	1278	1279	1278	1279	130050	130562
640	40960		5	1280	1281	1280	1281	1280	1281	1280	1281	131074	131586
767	49088			1534	1535	1534	1535	1534	1535	1534	1535	261122	261634
768	49152	3	6	1536	1537	1537	1536	1537	1536	1536	1537	3	515
895	57280			1790	1791	1791	1790	1791	1790	1790	1791	130051	130563
896	57344		7	1792	1793	1793	1792	1793	1792	1792	1793	131075	131587
1023	65472			2046	2047	2047	2046	2047	2046	2046	2047	261123	261635
130816	8372224	511	1022	261632	261633	261633	261632	261633	261632	261632	261633		
***	***												
130943	8380352			261886	261887	261887	261886	261887	261886	261886	261887		
130944	8380416		1023	261888	261889	261889	261888	261889	261888	261888	261889		
***	***												
131071	8388544			262142	262143	262143	262142	262143	262142	262142	262143		

Figure 4.11: MMUs Data-Flow

program. The table is at the subNTT granularity and distinguishes between the left and right NTTC sides. It doesn't however detail individual MMU lanes. Note that the right-most column in the table is the subNTT order resulting in the output, where the second to last column is used to index the data movement internally.

An interesting outcome of the wiring (transposing) is that the reading order between stages 1 and 2 requires Groups that are constructed from all even or all odd subNTTs. For instance, the first Group consists of subNTTs $\{0,512,1024,...\}$. By default, since they are even, these subNTTs are all on the left NTTC side. Doing nothing would result in a 50% utilization for stage 1 processing since only a single side can be used concurrently. To prevent this from happening, we group flipped the data between the left and right sides for odd Groups, as is marked by bold numbers in the table. The hardware implication of this is insignificant and it is easily handled by T16.

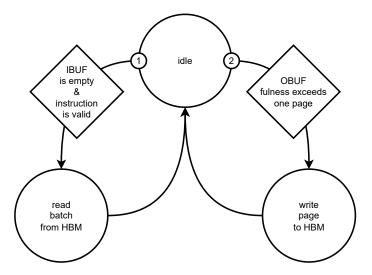


Figure 4.12: MMU State-Machine

4.4.3 T16 (REMAP-SS)

The purpose of the T16 module is to transpose the data in accordance with the CT FFT scheme from Figure 4.5. Achieving this along with the constraint of providing whole HBM pages at the output OBUFs leads to the minimal required input size of a Slice. The reason for requiring that T16 be able to handle two consecutive Slices (i.e. one Group) is due to the in-place requirements of the data being written back to the HBM. As it happens, the first Slice always transposes to page 0 of all output subNTTs in the Group, and the second Slice to page 1. This means that writing back the first Slice's output before reading the second Slice from the HBM would lead to writing before reading, a violation of the in-place assumption, and corruption of the data.

Since the storage requirements for T16's TBUFs are already high (i.e. a whole Group), the challenge is to suffice with that and read directly from there to the OBUFs (i.e. without requiring additional intermediate memory). The difficulty is that this requires storing data from all 16 lanes in their arrival order and organizing it into 16 TBUFs such that the relevant data to be read from the 16 TBUFs to the OBUFs is all exposed (i.e. that all

required samples for the next read are stored on mutually exclusive TBUFs). As it happens, this is possible but requires input and output staggering of the data within the TBUFs. The three columns of the table in Figure 4.13 show the organization of the input data, the data as it is stored on the TBUFs, and the output data. To understand the table, examine the first sample, marked 0, of the first four subNTTs on the left and the first four subNTTs on the right, in the input column. Clearly, making a single output row from them is not possible in their input arrangement, since each four occupy a single MMU lane. To enable this, we use a rotating mux to stagger the data as it enters the TBUFs, as can be seen in the middle column. Finally, when reading, all 0 marked samples can be read concurrently by simply manipulating the TBUFs addresses since they are already present on different lanes.

Other functions performed by T16 are the handling of data-flipping between the left and right NTTC sides to expose the required subNTTs for stage 1, as discussed previously, and a bypass function that allows skipping the transpose function at the final stage.

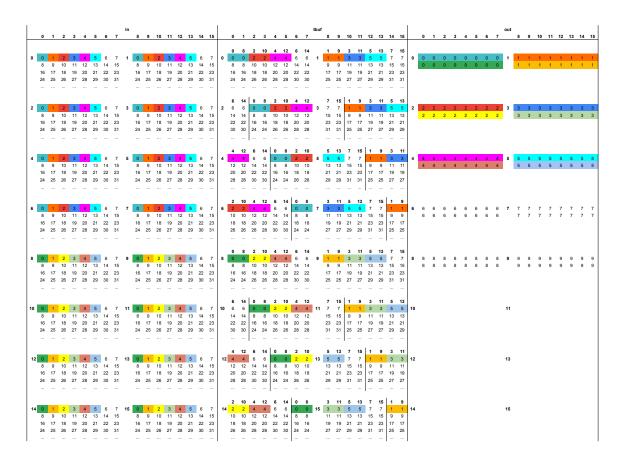


Figure 4.13: T16 Data Manipulation

4.4.4 NTTC (NTTC-SS)

NTTC is the core processor of the NTT design. In its heart, it relies on a radix-8 processor. Figure 4.14 illustrates the CT radix-8 implementation. This implementation requires

three radix-2 stages consisting of 12 butterflies and five non-trivial internal twiddle-factors. Note that the internal twiddle-factors are constant. The external twiddle-factors, used to construct larger NTTs based on this radix-8 core are handled via the input multipliers m0 to m7.

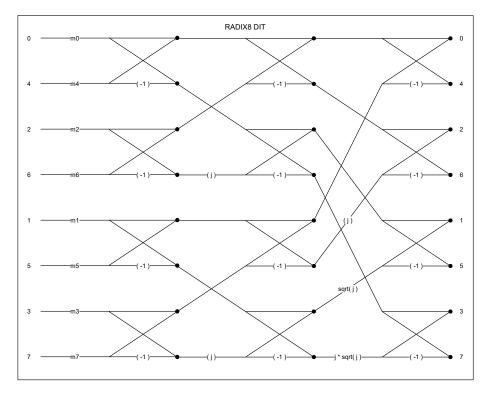


Figure 4.14: Radix-8 Winograd

To optimize the implementation we chose a Winograd-type implementation for the circuit based on [35]. This function-equivalent circuit is presented in Figure 4.15. Besides requiring only four non-trivial internal twiddle-factors, this implementation has a shorter critical path and is significantly more delay-balanced.

For maximum flexibility in constructing larger NTTs, we needed to understand how to use the radix-8 core in three additional operating points, radix-4, radix-2, and multiply-only. The prior two are necessary in order to construct NTTs whose length is not a power of eight³. The multiply-only option is used when external twiddle-factors, that are unavailable in memory, need to be constructed from the existing tables using multiplication (more on this soon). Radix-2 and multiply-only can be easily supported by exiting the pipeline prematurely. However, in the Winograd implementation, radix-4 does not seem to have a convenient exiting point⁴. To mend this we manipulated the Winograd circuit and came up with the alternative one as presented in Figure 4.16.

The main difference is the exchange of multiplication by -j for multiplication by j and other small rewiring changes required to maintain the original functionality. Note how the new circuit has one radix-4 available trivially at the top half and a second radix-4

³Our assumption is that all lengths of interest are a power-of-two.

⁴Note that we would like to support the following modes: a single radix-8, two parallel radix-4, or four parallel radix-2

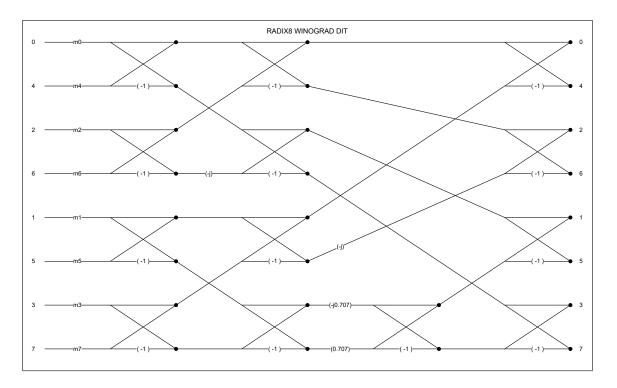


Figure 4.15: Radix-8 Winograd

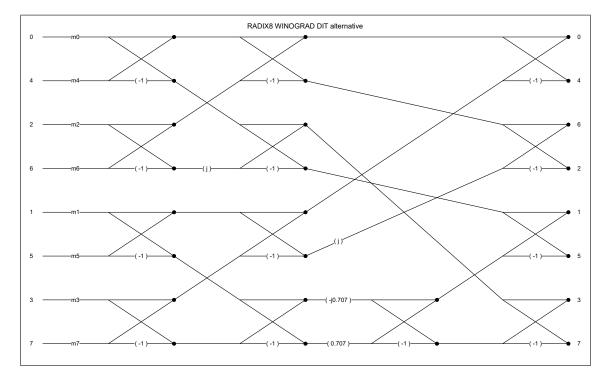


Figure 4.16: Radix-8 Winograd Alternative

available by multiplexing smartly to use the second j multiplier. The resulting fully-

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pipelined hardware circuit is depicted in Figure 4.17. Note that the circuit delay is constant for all modes.

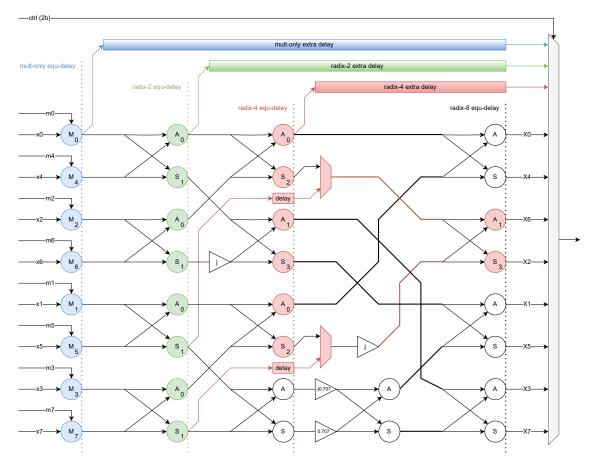


Figure 4.17: Radix-8 Circuit

The NTTC core is constructed around the radix-8 processor to perform subNTT processing over a Batch of subNTTs. Since NTTC works on a single side (left or right MMUs), the number of subNTTs it processes in one Batch is eight. The processing follows a CT scheme, in a similar manner to the full NTT scheme, but with out-of-place memory. We chose to allocate double the space for the intermediate NTTC memory, SBUF, in order to simplify the design and easily enable a zero-stall machine. Additionally, processing a Batch of subNTTs allowed the loosening of the CT dependencies between stages and permitted working with long latencies for the radix-8 engine. Figure 4.18 illustrates the general structure of the NTTC engine along with the input and output muxes for operating it for both the left and right NTTC sides. Note that the NTTC includes its own transpose-8 module T8 at the input to the SBUFs. The T8 module is responsible for the data transposition between stages and is built to support all previously discussed modes supported by the radix-8 processor.

Supporting all required external twiddle-factors was achieved by a three-level memory structure. The smallest required root-of-unity is of order 2²⁷. Storing all precalculated powers of this root-of-unity is wasteful and practically impossible. Simple analysis shows that the NTTC needs a table of 512 powers of the 512th root-of-unity powers for its

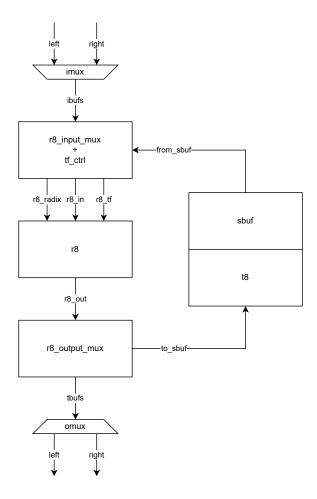


Figure 4.18: NTTC Architecture

subNTT processing. For the construction of the larger NTT, it requires the powers of roots-of-unity of orders 2^{18} and 2^{27} for stages 1 and 2, respectively. The required powers can be constructed from smaller tables by simple manipulation. Assume ω is a 2^{27} root-of-unity and we need the twiddle-factor $\omega^{\rho l}$ in stage 2, where $\rho \in [0, 2^{18})$ and $l \in [0, 2^9)$ (see Section 1.4 for more detail). We can multiply the two integers $r = \rho l$ resulting in a 27-bit integer that can be represented by three 9-bit digits $r = a2^{18} + b2^9 + c$. The required twiddle-factor can thus be represented as $\omega^r = (\omega^{2^{18}})^a \cdot (\omega^{2^9})^b \cdot \omega^c$. It is easy to see that $\omega^{2^{18}}$, ω^{2^9} are the roots-of-unity of orders 2^9 and 2^{18} , respectively, and since a, b, and c are 9-bit digits, all required twiddle-factors can be constructed using three 512-entry look-up tables. An illustration of the NTTC, detailing its memory structure is presented in Figure 4.19

4.4.5 Memory Estimation

Given the planned architecture, as presented in this section, we estimated the memory usage of the design. To simplify the placement and routing of the design, along with previous constraints like limiting NTTC-SS to a single SLR, we attempted to use no more than 50% of the existing memory resources. The largest required memory is in REMAP-SS

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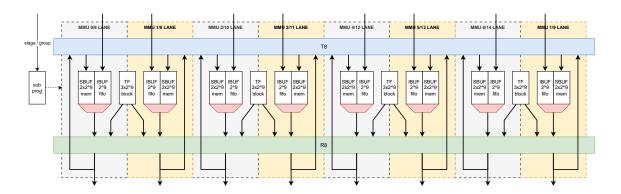


Figure 4.19: NTTC Structure

and we chose to use URAMs for it. The rest of the memories were defined to use BRAMs. The table in Figure 4.20 details our estimation.

	c1100	usage
bram (36Kb)	1344	19.05%
uram (288Kb)	640	40.00%
hbm (GB)	8	100.00%

	quant	usage	remarks
dp bram width	bits	32	w/o parity
dp bram depth	#	1024	
bram width	bits	64	w/o parity
bram depth	#	512	
uram width	bits	64	w/o parity
uram depth	#	4096	
nof nttc	#	1	
nof virtual nttc (vnttc)	#	2	
nof mmu / vnttc	#	8	
nof stages	#	3	
Word size	bits	256	
subntt size	Words	512	
batch size	subntts	8	
batch size	Words	4096	
batch size / mmu / vnttc	Words	512	
slice size	batches	16	
slice size	Words	65536	
slice size / mmu / vnttc	Words	8192	
tf table size	Words	1536	
ibuf / mmu / vnttc	brams	4	fifo
obuf / mmu / vnttc	brams	4	fifo
sbuf / mmu / nttc	brams	8	double buffer
tf / mmu / nttc	brams	8	dual read port
tbuf / mmu / vnttc	urams	16	double buffer
ibuf	brams	64	
obuf	brams	64	
sbuf	brams	64	
tf	brams	64	
tbuf	urams	256	
total urams	urams	256	
total brams	brams	256	

Figure 4.20: Memory Estimation

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