

GPU Implementation of Data-Aided Equalizers

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Master of Science

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ABSTRACT

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Multipath is one of the dominant causes for link loss in aeronautical telemetry. Equalizers have been studied to combat multipath interference in aeronautical telemetry. Blind Constant Modulus Algorithm (CMA) equalizers are currently being used on SOQPSK-TG. The Preamble Assisted Equalization (PAQ) has been funded by the Air Force to study data-aided equalizers on SOQPSK-TG. PAQ compares side by side no equalization, data-aided zero forcing equalization, data-aided MMSE equalization, data-aided initialized CMA equalization, data-aided frequency domain equalization, and blind CMA equalization. An real time experimental test setup has been assembled including an RF receiver for data acquisition, FPGA for hardware interfacing and buffering, GPUs for signal processing, spectrum analyzer for viewing multipath events, and an 8 channel bit error rate tester to compare equalization performance. Lab tests were done with channel and noise emulators. Flight tests were conducted in March 2016 and June 2016 at Edwards Air Force Base to test the equalizers on live signals. The test setup achieved a 10Mbps throughput with a 6 second delay. Counter intuitive to the simulation results, the flight tests at Edwards AFB in March and June showed blind equalization is superior to data-aided equalization. Lab tests revealed some types of multipath caused timing loops in the RF receiver to produce garbage samples. Data-aided equalizers based on data-aided channel estimation leads to high bit error rates. A new experimental setup is been proposed, replacing the RF receiver with a RF data acquisition card. The data acquisition card will always provide good samples because the card has no timing loops, regardless of severe multipath.

Keywords: MISSING

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Students may use the acknowledgments page to express appreciation for the committee members, friends, or family who provided assistance in research, writing, or technical aspects of the dissertation, thesis, or selected project. Acknowledgments should be simple and in good taste.

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Chapter 1

Introduction

This is the introduction

Chapter 2

Problem Statement

This is the Problem Statement

Some algorithms map very well to CPUs because they are computationally light, but what happens why your CPU cannot achieve the desired throughput or data rate?

In the past, the answer was FPGAs. But now, with Graphics Processing Units getting bigger faster stronger, there has been a recent pull towards GPUs because of

the ease of implementation vs HDL programming

the ease of setup

Chapter 3

Signal Processing with GPUs

This thesis explores the use of GPUs in data-aided estimation, equalization and filtering operations.

A Graphics Processing Unit (GPU) is a computational unit with a highly-parallel architecture well-suited for executing the same function on many data elements. In the past, GPUs were used to process graphics data. Recently, general purpose GPUs are being used for high performance computing in computer vision, deep learning, artificial intelligence and signal processing [1].

GPUs cannot be programmed the way as a CPU. NVIDIA released a extension to C, C++ and Fortran called CUDA (Compute Unified Device Architecture). CUDA allows a programmer to write C++ like functions that are massively parallel called *kernels*. To invoke parallelism, a GPU kernel is called N times and mapped to N *threads* that run concurrently. To achieve the full potential of high performance GPUs, kernels must be written with some basic concepts about GPU architecture and memory in mind.

The purpose of this overview is to provide context for the contributions of this thesis. As such this overview is not a tutorial. For a full explanation of CUDA programming please see the CUDA toolkit documentation [3].

3.1 Simple GPU code example

If a programmer has some C++ experience, learning how to program GPUs using CUDA comes fairly easily. GPU code still runs top to bottom and memory still has to be allocated. The only real difference is where the memory physically is and how functions run on GPUs. To run functions or kernels on GPUs, the memory must be copied from the host (CPU) to the device (GPU). Once the memory has been copied, the parallel GPU kernels can be called. After the GPU

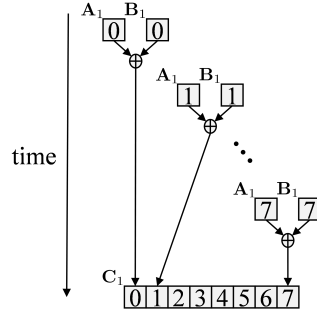


Figure 3.1: A block diagram of how a CPU sequentially performs vector addition.

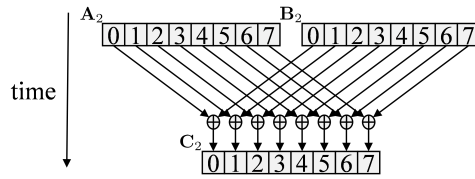


Figure 3.2: A block diagram of how a GPU performs vector addition in parallel.

kernels have finished, the resulting memory has to be copied back from the device (GPU) to the host (CPU).

Listing 3.1 shows a simple program that sums to vectors together

$$\begin{aligned} \mathbf{C}_1 &= \mathbf{A}_1 + \mathbf{B}_1 \\ \mathbf{C}_2 &= \mathbf{A}_2 + \mathbf{B}_2 \end{aligned} \tag{3.1}$$

where each vector is length 1024. Figure 3.1 shows how the CPU computes \mathbf{C}_1 by summing elements of \mathbf{A}_1 and \mathbf{B}_1 together *sequentially*. Figure 3.2 shows how the GPU computes \mathbf{C}_2 by summing elements of \mathbf{A}_1 and \mathbf{B}_1 together *in parallel*. The GPU kernel computes every element of \mathbf{C}_2 in parallel while the CPU computes one element of \mathbf{C}_1 at a time.

Listing 3.1: Comparison of CPU verse GPU code.

```
1  #include <iostream>
2  #include <stdlib.h>
3  #include <math.h>
4  using namespace std;
5
6  void VecAddCPU(int* destination, int* source0, int* source1, int myLength) {
7      for(int i = 0; i < myLength; i++)
8          destination[i] = source0[i] + source1[i];
9  }
10
11 __global__ void VecAddGPU(int* destination, int* source0, int* source1, int lastThread) {
12     int i = blockIdx.x*blockDim.x + threadIdx.x;
13
14     // don't access elements out of bounds
15     if(i >= lastThread)
16         return;
17
18     destination[i] = source0[i] + source1[i];
19 }
20
21 int main() {
22     int numPoints = pow(2,22);
23     cout << numPoints << endl;
24     /*-----
25                                     CPU Start
26     -----*/
27     // allocate memory on host
28     int *A1;
29     int *B1;
30     int *C1;
31     A1 = (int*) malloc (numPoints*sizeof(int));
32     B1 = (int*) malloc (numPoints*sizeof(int));
33     C1 = (int*) malloc (numPoints*sizeof(int));
34
35     // Initialize vectors 0-99
36     for(int i = 0; i < numPoints; i++){
37         A1[i] = rand()%100;
38         B1[i] = rand()%100;
39     }
40
41     // vector sum C1 = A1 + B1
42     VecAddCPU(C1, A1, B1, numPoints);
43     /*-----
44                                     CPU End
45     -----*/
46
47     /*-----
48                                     GPU End
49     -----*/
50     // allocate memory on host for result
51     int *C2;
52     C2 = (int*) malloc (numPoints*sizeof(int));
53
54     // allocate memory on device for computation
55     int *A2_gpu;
56     int *B2_gpu;
57     int *C2_gpu;
58     cudaMalloc(&A2_gpu, sizeof(int)*numPoints);
59     cudaMalloc(&B2_gpu, sizeof(int)*numPoints);
60     cudaMalloc(&C2_gpu, sizeof(int)*numPoints);
61
62     // Copy vectors A and B from host to device
63     cudaMemcpy(A2_gpu, A1, sizeof(int)*numPoints, cudaMemcpyHostToDevice);
64     cudaMemcpy(B2_gpu, B1, sizeof(int)*numPoints, cudaMemcpyHostToDevice);
65
66     // Set optimal number of threads per block
```

```

67     int numTreadsPerBlock = 32;
68
69     // Compute number of blocks for set number of threads
70     int numBlocks = numPoints/numTreadsPerBlock;
71
72     // If there are left over points, run an extra block
73     if(numPoints % numTreadsPerBlock > 0)
74         numBlocks++;
75
76     // Run computation on device
77     //for(int i = 0; i < 100; i++)
78         VecAddGPU<<<numBlocks, numTreadsPerBlock>>>>(C2_gpu, A2_gpu, B2_gpu, numPoints);
79
80     // Copy vector C2 from device to host
81     cudaMemcpy(C2, C2_gpu, sizeof(int)*numPoints, cudaMemcpyDeviceToHost);
82     /*-----
83                                     GPU End
84     -----*/
85
86     // Free vectors on CPU
87     free(A1);
88     free(B1);
89     free(C1);
90     free(C2);
91
92     // Free vectors on GPU
93     cudaFree(A2_gpu);
94     cudaFree(B2_gpu);
95     cudaFree(C2_gpu);
96 }

```

3.2 GPU kernel using threads and thread blocks

A GPU kernel is executed on a GPU by launching $\text{numTreadsPerBlock} \times \text{numBlocks}$ threads. Each thread has a unique index. CUDA calls this index `threadIdx` and `blockIdx`. `threadIdx` is the thread index inside the assigned thread block. `blockIdx` is the index of the block the thread is assigned. `blockDim` is the number of threads assigned per block, in fact `blockDim = numTreadsPerBlock`. Both `threadIdx` and `blockIdx` are three dimensional and have x, y and z components. In this thesis only the x dimension is used because GPU kernels operate only on vectors.

To replace a CPU for loop that runs 0 to $N - 1$, a GPU kernel launches N threads with T threads per thread block. The number of blocks needed is $M = \frac{N}{T}$ or $M = \frac{N}{T} + 1$ if N is not an integer multiple of T . Figure 3.3 shows 32 threads launched in 4 thread blocks with 8 threads per block. Figure 3.4 shows 36 threads launched in 5 thread blocks with 8 threads per block. An full extra thread block must be launched to with 8 threads but 4 threads are idle.

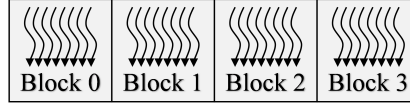


Figure 3.3: Block 0 32 threads launched in 4 thread blocks with 8 threads per block.

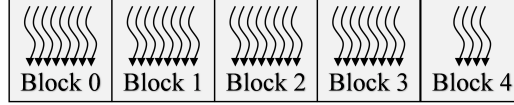


Figure 3.4: 36 threads launched in 5 thread blocks with 8 threads per block with 4 idle threads.

3.3 GPU memory

Thread blocks run independent of other thread blocks. The GPU does not guarantee Block 0 will execute before Block 2. Threads in blocks can coordinate and use shared memory but blocks do not coordinate with other blocks. Threads have access to private local memory that is fast and efficient. Each thread in a thread block has access to private shared memory in the thread block. All threads have access to global memory.

Local memory is the fastest and global memory is by far the slowest. One global memory access takes 400-800 clock cycles while a local memory is a few clock cycles. Why not just do all computations in local memory? The memory needs come from global memory to before it can be used in local memory. Memory should be saved in shared memory if many threads are going to use it in a thread block. Local and shared memory should be used as much as possible but sometimes a GPU kernel cant utilized local and shared memory because elements might only be used once.

Why is global memory so slow? Looking at the physical hardware will shed some light. This thesis uses NVIDIA Tesla K40c and K20c GPUs, Table 3.1 gives some specifications and Figure 3.6 shows the form factor of the these GPUs. The red box in Figure 3.7 show the GPU chip and the yellow boxes show the SRAM that is *off* the GPU chip. The GPU global memory is located in the SRAM. To move memory to thread blocks *on* the GPU chip from global memory

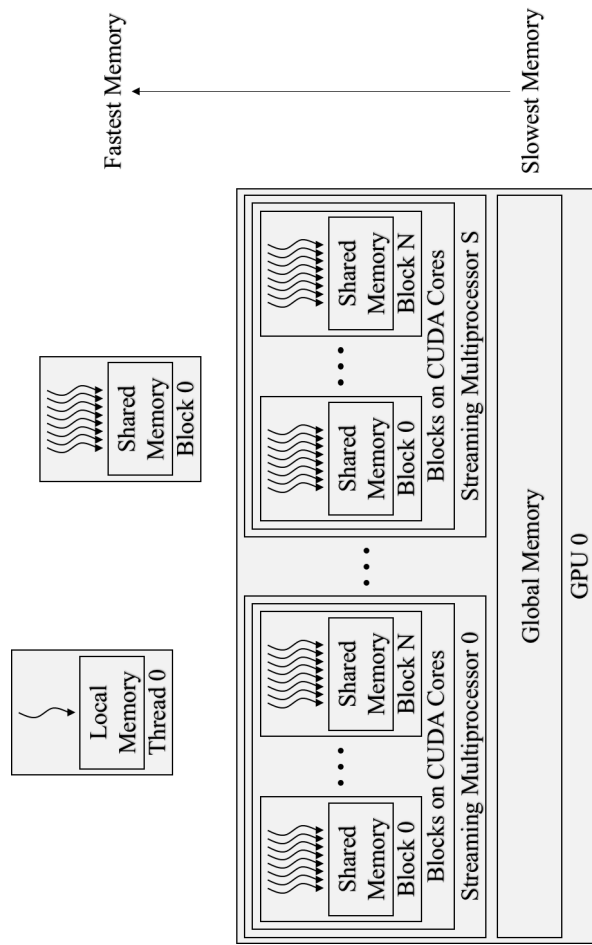


Figure 3.5: A block diagram where local, shared, and global memory is located. Each thread has private local memory. Each thread block has private shared memory. The GPU has global memory that all threads can access.



Figure 3.6: NVIDIA Tesla K40c and K20c.

Feature	Tesla K40c	Tesla K20c
Memory size (GDDR5)	12 GB	5 GB
CUDA cores	2880	2496
Base clock (MHz)	745	732

Table 3.1: The computational resources available with three NVIDIA GPUs used in this thesis (1x Tesla K40c 2x Tesla K20c).

requires fetching memory from *off* the GPU. Now 400-800 clock cycles doesn't sound all the bad huh?

3.4 Cuda Libraries

CUDA isn't just a programming language, it also has many libraries that are extremely useful that are optimized for NVIDIA GPUs. CUDA libraries are written by NVIDIA engineers

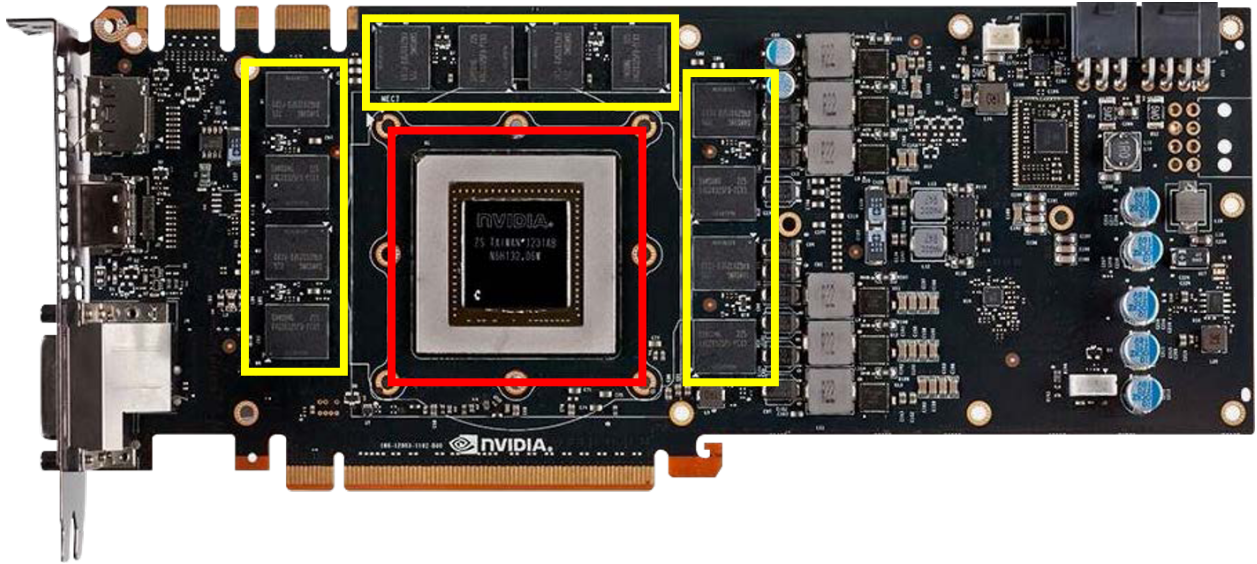


Figure 3.7: Example of an NVIDIA GPU card. The SRAM is shown to be boxed in yellow. The GPU chip is shown to be boxed in red.

that know how to squeeze out every drop of performance out of NVIDIA GPUs. Because ninjas are unbeatable, NVIDIA engineers are known as ninjas in the Telemetry Group at BYU. While figuring out how to optimize a GPU kernel is extremely satisfying, GPU programmers should always search the CUDA libraries for any thing that might be useful.

Some libraries used in this Thesis are

- cufft
- cublas
- cusolver
- cusolverSp

3.5 Cuda Convolution

An important tool to Digital Communications and Digital Signal Processing is convolution. Time domain convolution is

$$y(n) = \sum_{m=0}^{L_h-1} x(m)h(n-m) \quad (3.2)$$

where $x(m)$, $h(n-m)$ and $y(n)$ are complex. While Listing 3.1 is a simple and good example showing how program GPUs, frankly, it is pretty boring and doesn't display the real challenges and tradeoffs of GPUs. Listing 3.2 shows four different ways of implementing convolution

- time domain convolution in a CPU
- time domain convolution in a GPU using global memory
- time domain convolution in a GPU using shared memory
- frequency domain convolution in a GPU using CUDA libraries

The CPU implements Equation (3.2) in ConvCPU directly from line 6 to 30. The GPU implements time domain convolution using global memory in the GPU kernel ConvGPU. ConvGPU on lines 32 to 59 is a parallel of ConvCPU. ConvGPU implements time domain convolution by accessing global memory for any needed element of the signal and filter.

Threads accessing the same elements of the filter in global memory seems like a waste of valuable clock cycles. The GPU kernel ConvGPUshared on lines 61 to 96 differs slightly from ConvGPU by making L_h threads save the filter to shared memory from global memory.

Any graduate of a signal processing class knows the trade off of convolution in the time verse frequency domain. Time domain convolution takes $\approx L_s L_h$ complex multiplies where L_s is the signal length and L_h is the filter length. Frequency domain convolution takes $\approx \frac{3N_{\text{FFT}}}{2} \log_2(N_{\text{FFT}}) + N_{\text{FFT}}$ complex multiplies where N_{FFT} is the next multiple of 2 above the convolution length $L_s + L_h - 1$.

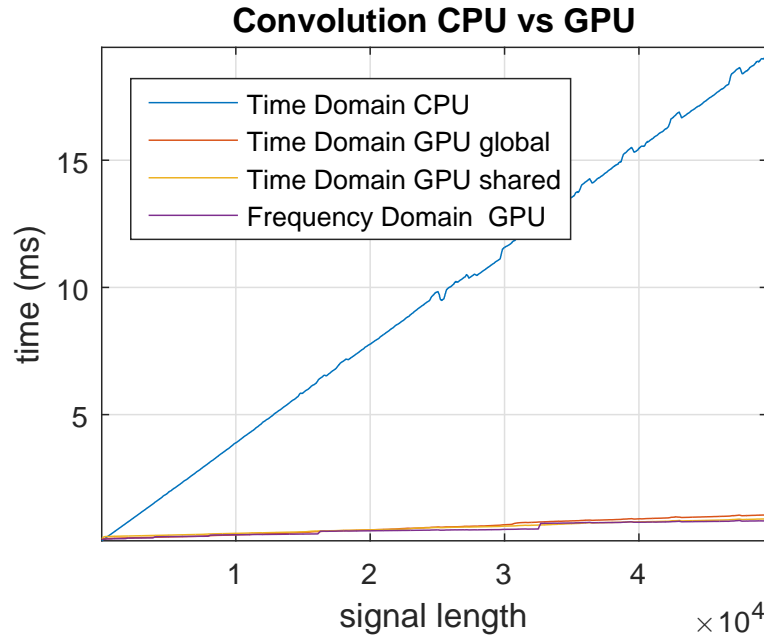


Figure 3.8: Comparison of complex convolution in on CPU to GPU with varying signal lengths.

Lines 213 to 232 show how to do frequency domain convolution using the cuFFT library.

So the questions are: When should I stay on the host and run my convolution on the CPU? When should I copy my vectors to the GPU, run my convolution, then copy the vectors back to the host? If I am going to run my convolution the GPU, when should I only use global memory? When should I use shared memory? When should I do convolution in the frequency domain?

The answer to all of the questions is...it depends on your signal length, filter length, CPU, GPU and memory.

The CUDA programmer can make an educated guess on which way will be fastest, but until the computations have been done in the GPU there is no definite answer. Figure 3.8 shows a comparison of computation time by varying the signal length with a set filter length. The signal length is varied from 1 to 50000 and the filter length is 186; The functions in 3.1 where ran on a NVIDIA Tesla K40c GPU. Judging my the Figure, for signals of any significant length, GPU convolution is much faster than CPU.

Is it ever better to do convolution on the CPU rather than GPU? Yes, but for very short signals with a 186 tap filter. Figure 3.9 shows when a CPU is faster than a GPU.

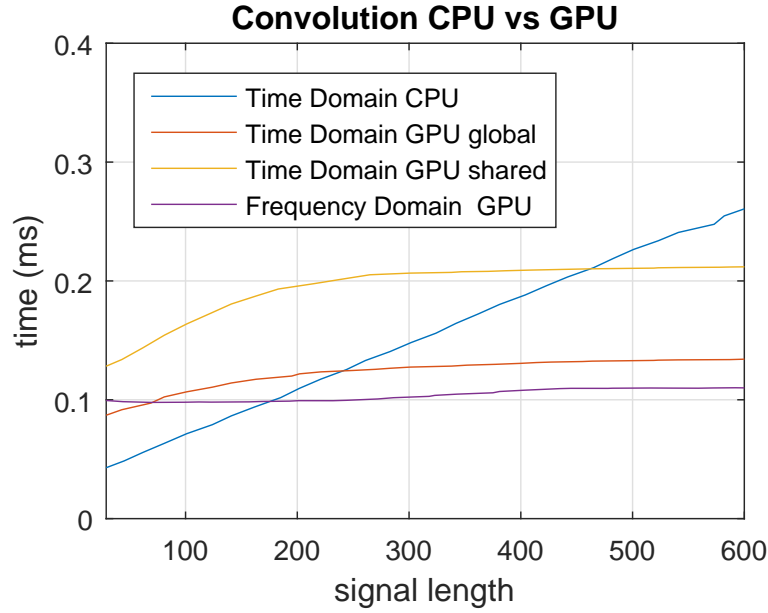


Figure 3.9: Plot showing when CPU convolution is faster than GPU convolution.

Now for the interesting question, if I am going to do convolution in the GPU, should I do time domain or frequency domain convolution? If I do time domain convolution, should I use global or shared memory? Figure 3.10 shows that the answer is, “it depends”. A good choice is to do frequency domain, but not always. As the signal length increases, the frequency domain execution time has steps because the signal is zero padded out to the next power of 2 to leverage the Cooley-Tukey FFT algorithm [?].

Usually, when implementing convolution, the signal and filter length is set. It is good practice to implement convolution in the CPU and GPU in many ways to evaluate which way is best. Choosing the signal length to be 2^{15} and the filter length to be 186, Table 3.2 shows how the different algorithms compare.

3.6 Thread Optimization

When first introduced to GPUs, a programmer might be tempted to launch as many threads per block possible. But, notice in Listing 3.2 lines 190, 203, 219 and 225 launch a minimum of 96 threads per block and maximum of 192 threads per block. Running the GPU at lower occupancy

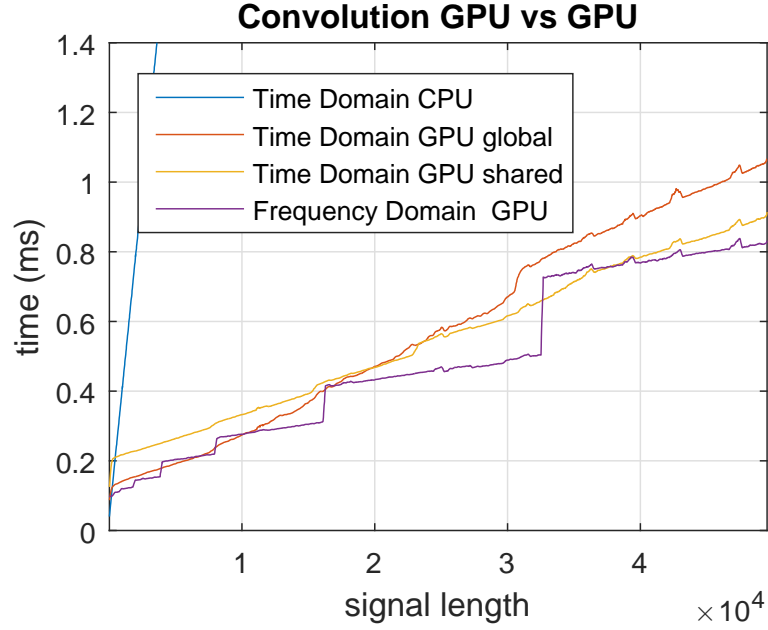


Figure 3.10: Plot showing trade offs with convolution in GPUs.

Algorithm	time (ms)
CPU time domain (ConvCPU)	131.943
GPU time domain global (ConvGPU)	7.79477
GPU time domain shared (ConvGPUshared)	6.77736
GPU frequency domain shared	5.81085

Table 3.2: Convolution computation times with signal length $2^{15} = 32768$ and filter length 186 on a Tesla K40c GPU.

leaves each thread with more resources and memory bandwidth. If 1024 threads per block were always launched, lighter computaion GPU kernels will be starving for memory while other thread blocks eat up the memory bandwidth.

Improving memory accesses should always be the first optimization when a GPU kernel needs to be faster. The next step is to find the optimal number of threads per block to launch. The number of threads per block affect the amount of resources available to each thread. If a kernel is more computationally heavy, launching less threads per block may lead to faster execution time because threads have more resources available.

Knowing the perfect number of threads per block to launch is challenging to calculate. Luckily, there is a finite number of possible threads per block, 1 to 1024. A simple test program could time a GPU kernel while sweeping the number of threads per block from 1 to 1024. The number of threads per block with the fastest computation time is the optimal number of threads per block for that specific GPU kernel.

Most of the time the optimal number of threads per block is a multiple of 32. At the lowest level of architecture, GPU do computations in *warps*. Warps are groups of 32 threads to do every computation together in lock step. If the number of threads per block is a non multiple of 32, some threads in a warp will be idle and the GPU will be wasting resources.

Figure ?? shows the execution time of ConvGPU while varying threads per block. The minimum execution time is 651651 at 128 threads per block.

Listing 3.2: CUDA code to performing complex convolution four different ways: time domain CPU, time domain GPU, time domain GPU using shared memory and frequency domain GPU.

```

1  #include <cuFFT.h>
2  using namespace std;
3  // Length of Filter (186 is a magic number)
4  const int LH = 186;
5
6  void ConvCPU(cuFFTComplex* y, cuFFTComplex* x, cuFFTComplex* h, int Lx, int Lh) {
7      for(int yIdx = 0; yIdx < Lx+Lh-1; yIdx++){
8          cuFFTComplex temp;
9          temp.x = 0;
10         temp.y = 0;
11         for(int hIdx = 0; hIdx < Lh; hIdx++){
12             int xAccessIdx = yIdx-hIdx;
13             if(xAccessIdx>=0 && xAccessIdx<Lx){
14                 // temp += x[xAccessIdx]*h[hIdx];
15                 // (A+jB)(C+jD) = (AC-BD) + j(AD+BC)
16                 float A = x[xAccessIdx].x;
17                 float B = x[xAccessIdx].y;
18                 float C = h[hIdx].x;
19                 float D = h[hIdx].y;
20                 cuFFTComplex complexMult;
21                 complexMult.x = A*C-B*D;
22                 complexMult.y = A*D+B*C;
23
24                 temp.x += complexMult.x;
25                 temp.y += complexMult.y;
26             }
27         }
28         y[yIdx] = temp;
29     }
30 }
31
32 __global__ void ConvGPU(cuFFTComplex* y, cuFFTComplex* x, cuFFTComplex* h, int Lx, int Lh) {
33     int yIdx = blockIdx.x*blockDim.x + threadIdx.x;
34     int lastThread = Lx+Lh-1;
35     // Don't access elements out of bounds
36     if(yIdx >= lastThread)
37         return;
38     cuFFTComplex temp;
39     temp.x = 0;
40     temp.y = 0;
41     for(int hIdx = 0; hIdx < Lh; hIdx++){
42         int xAccessIdx = yIdx-hIdx;
43         if(xAccessIdx>=0 && xAccessIdx<Lx){
44             // temp += x[xAccessIdx]*h[hIdx];
45             // (A+jB)(C+jD) = (AC-BD) + j(AD+BC)
46             float A = x[xAccessIdx].x;
47             float B = x[xAccessIdx].y;
48             float C = h[hIdx].x;
49             float D = h[hIdx].y;
50             cuFFTComplex complexMult;
51             complexMult.x = A*C-B*D;
52             complexMult.y = A*D+B*C;
53
54             temp.x += complexMult.x;
55             temp.y += complexMult.y;
56         }
57         y[yIdx] = temp;
58     }
59 }
60
61 __global__ void ConvGPUshared(cuFFTComplex* y, cuFFTComplex* x, cuFFTComplex* h, int Lx, int Lh) {
62     int yIdx = blockIdx.x*blockDim.x + threadIdx.x;
63     // Be sure to read in full h_shared before checking lastThread
64     // First Lh threads in thread block read h from global memory into shared memory
65     __shared__ cuFFTComplex h_shared[LH];

```

```

66     if(threadIdx.x < LH)
67         h_shared[threadIdx.x] = h[threadIdx.x];
68     int lastThread = Lx+Lh-1;
69     // Don't access elements out of bounds
70     if(yIdx >= lastThread)
71         return;
72     // Thread barrier, ensures threads wait until h has been read into h_shared
73     __syncthreads();
74     cufftComplex temp;
75     temp.x = 0;
76     temp.y = 0;
77     for(int hIdx = 0; hIdx < Lh; hIdx++){
78         int xAccessIdx = yIdx-hIdx;
79         if(xAccessIdx>=0 && xAccessIdx<Lx){
80             // temp += x[xAccessIdx]*h_shared[hIdx];
81
82             // (A+jB)(C+jD) = (AC-BD) + j(AD+BC)
83             float A = x[xAccessIdx].x;
84             float B = x[xAccessIdx].y;
85             float C = h_shared[hIdx].x;
86             float D = h_shared[hIdx].y;
87             cufftComplex complexMult;
88             complexMult.x = A*C-B*D;
89             complexMult.y = A*D+B*C;
90
91             temp.x += complexMult.x;
92             temp.y += complexMult.y;
93         }
94         y[yIdx] = temp;
95     }
96 }
97
98 __global__ void PointToPointMultiply(cufftComplex* vec0, cufftComplex* vec1, int lastThread){
99     int i = blockIdx.x*blockDim.x + threadIdx.x;
100    // Don't access elements out of bounds
101    if(i >= lastThread)
102        return;
103    // vec0[i] = vec0[i]*vec1[i];
104    // (A+jB)(C+jD) = (AC-BD) + j(AD+BC)
105    float A = vec0[i].x;
106    float B = vec0[i].y;
107    float C = vec1[i].x;
108    float D = vec1[i].y;
109    cufftComplex complexMult;
110    complexMult.x = A*C-B*D;
111    complexMult.y = A*D+B*C;
112    vec0[i] = complexMult;
113 }
114
115 __global__ void ScalarMultiply(cufftComplex* vec0, float scalar, int lastThread){
116     int i = blockIdx.x*blockDim.x + threadIdx.x;
117     // Don't access elements out of bounds
118     if(i >= lastThread)
119         return;
120     cufftComplex scalarMult;
121     scalarMult.x = vec0[i].x*scalar;
122     scalarMult.y = vec0[i].y*scalar;
123     vec0[i] = scalarMult;
124 }
125
126 int main(){
127     int mySignalLength = pow(2,15);
128     int myFilterLength = LH;
129     int myConvLength = mySignalLength + myFilterLength - 1;
130     int Nfft = pow(2, ceil(log(myConvLength)/log(2)));
131     int numTreadsPerBlock;
132     int numBlocks;
133 }

```

```

134     cufftHandle plan;
135     int n[1] = {Nfft};
136     cufftPlanMany(&plan, 1, n, NULL, 1, 1, NULL, 1, 1, CUFFT_C2C, 1);
137
138     // Allocate memory on host
139     cufftComplex *mySignal1;
140     cufftComplex *myFilter1;
141     cufftComplex *myConv1;
142     cufftComplex *myConv2;
143     cufftComplex *myConv3;
144     cufftComplex *myConv4;
145     mySignal1 = (cufftComplex*) malloc(mySignalLength*sizeof(cufftComplex));
146     myFilter1 = (cufftComplex*) malloc(myFilterLength*sizeof(cufftComplex));
147     myConv1 = (cufftComplex*) malloc(myConvLength * sizeof(cufftComplex));
148     myConv2 = (cufftComplex*) malloc(myConvLength * sizeof(cufftComplex));
149     myConv3 = (cufftComplex*) malloc(myConvLength * sizeof(cufftComplex));
150     myConv4 = (cufftComplex*) malloc(Nfft * sizeof(cufftComplex));
151     for(int i = 0; i < mySignalLength; i++){
152         mySignal1[i].x = rand()%100-50;
153         mySignal1[i].y = rand()%100-50;
154     }
155     for(int i = 0; i < myFilterLength; i++){
156         myFilter1[i].x = rand()%100-50;
157         myFilter1[i].y = rand()%100-50;
158     }
159
160     // Allocate memory on device
161     cufftComplex *dev_mySignal2;
162     cufftComplex *dev_myFilter2;
163     cufftComplex *dev_myConv2;
164     cufftComplex *dev_mySignal3;
165     cufftComplex *dev_myFilter3;
166     cufftComplex *dev_myConv3;
167     cufftComplex *dev_mySignal4;
168     cufftComplex *dev_myFilter4;
169     cufftComplex *dev_myConv4;
170     cudaMalloc(&dev_mySignal2, mySignalLength*sizeof(cufftComplex));
171     cudaMalloc(&dev_myFilter2, myFilterLength*sizeof(cufftComplex));
172     cudaMalloc(&dev_myConv2, myConvLength * sizeof(cufftComplex));
173     cudaMalloc(&dev_mySignal3, mySignalLength*sizeof(cufftComplex));
174     cudaMalloc(&dev_myFilter3, myFilterLength*sizeof(cufftComplex));
175     cudaMalloc(&dev_myConv3, myConvLength * sizeof(cufftComplex));
176     cudaMalloc(&dev_mySignal4, Nfft * sizeof(cufftComplex));
177     cudaMalloc(&dev_myFilter4, Nfft * sizeof(cufftComplex));
178     cudaMalloc(&dev_myConv4, Nfft * sizeof(cufftComplex));
179
180     /*-----
181         CPU Time Domain Direct Convolution
182     -----*/
183     ConvCPU(myConv1, mySignal1, myFilter1, mySignalLength, myFilterLength);
184
185     /*-----
186         GPU Time Domain Direct Convolution
187     -----*/
188     cudaMemcpy(dev_mySignal2, mySignal1, sizeof(cufftComplex)*mySignalLength,
189               cudaMemcpyHostToDevice);
189     cudaMemcpy(dev_myFilter2, myFilter1, sizeof(cufftComplex)*myFilterLength,
190               cudaMemcpyHostToDevice);
191     numTreadsPerBlock = 128;
192     numBlocks = myConvLength/numTreadsPerBlock;
193     if(myConvLength % numTreadsPerBlock > 0)
194         numBlocks++;
195     ConvGPU(<<numBlocks, numTreadsPerBlock>>>(&dev_myConv2, dev_mySignal2, dev_myFilter2,
196         mySignalLength, myFilterLength);
197     cudaMemcpy(myConv2, dev_myConv2, myConvLength*sizeof(cufftComplex),
198               cudaMemcpyDeviceToHost);

```



```

198  /*-----
199  GPU Time Domain Convolution Using Shared Memory
200  -----*/
201  cudaMemcpy(dev_mySignal3, mySignal1, sizeof(cufftComplex)*mySignalLength,
202             cudaMemcpyHostToDevice);
203  cudaMemcpy(dev_myFilter3, myFilter1, sizeof(cufftComplex)*myFilterLength,
204             cudaMemcpyHostToDevice);
205  numTreadsPerBlock = 192;
206  numBlocks = myConvLength/numTreadsPerBlock;
207  if(myConvLength % numTreadsPerBlock > 0)
208      numBlocks++;
209  ConvGPUshared<<<numBlocks, numTreadsPerBlock>>>(dev_myConv3, dev_mySignal3, dev_myFilter3
210      , mySignalLength, myFilterLength);
211  cudaMemcpy(myConv3, dev_myConv3, myConvLength*sizeof(cufftComplex),
212             cudaMemcpyDeviceToHost);
213
214  /*-----
215  GPU Frequency Domain Convolution using cuFFT
216  -----*/
217  cudaMemset(dev_mySignal4, 0, Nfft*sizeof(cufftComplex));
218  cudaMemset(dev_myFilter4, 0, Nfft*sizeof(cufftComplex));
219  cudaMemcpy(dev_mySignal4, mySignal1, sizeof(cufftComplex)*mySignalLength,
220             cudaMemcpyHostToDevice);
221  cudaMemcpy(dev_myFilter4, myFilter1, sizeof(cufftComplex)*myFilterLength,
222             cudaMemcpyHostToDevice);
223  cufftExecC2C(plan, dev_mySignal4, dev_mySignal4, CUFFT_FORWARD);
224  cufftExecC2C(plan, dev_myFilter4, dev_myFilter4, CUFFT_FORWARD);
225  numTreadsPerBlock = 96;
226  numBlocks = Nfft/numTreadsPerBlock;
227  if(Nfft % numTreadsPerBlock > 0)
228      numBlocks++;
229  PointToPointMultiply<<<numBlocks, numTreadsPerBlock>>>(dev_mySignal4, dev_myFilter4, Nfft
230      );
231  cufftExecC2C(plan, dev_mySignal4, dev_mySignal4, CUFFT_INVERSE);
232  numTreadsPerBlock = 128;
233  numBlocks = Nfft/numTreadsPerBlock;
234  if(Nfft % numTreadsPerBlock > 0)
235      numBlocks++;
236  float scalar = 1.0/((float)Nfft);
237  ScalarMultiply<<<numBlocks, numTreadsPerBlock>>>(dev_mySignal4, scalar, Nfft);
238  cudaMemcpy(myConv4, dev_mySignal4, myConvLength*sizeof(cufftComplex),
239             cudaMemcpyDeviceToHost);
240  cufftDestroy(plan);
241
242  // Free vectors on CPU
243  free(mySignal1);
244  free(myFilter1);
245  free(myConv1);
246  free(myConv2);
247  free(myConv3);
248  free(myConv4);
249
250  // Free vectors on GPU
251  cudaFree(dev_mySignal2);
252  cudaFree(dev_myFilter2);
253  cudaFree(dev_myConv2);
254  cudaFree(dev_mySignal2);
255  cudaFree(dev_myFilter2);
256  cudaFree(dev_myConv2);
257  cudaFree(dev_mySignal3);
258  cudaFree(dev_myFilter3);
259  cudaFree(dev_myConv3);
260  cudaFree(dev_mySignal4);
261  cudaFree(dev_myFilter4);
262  cudaFree(dev_myConv4);
263
264  return 0;
265 }

```


Chapter 4

System Overview

4.1 Overview

This chapter gives a high lever overview of the the algorithms implemented into the GPU. A block diagram is shown in Figure 4.1. The algorithms implemented in GPUs will briefly be explained. Chapter 5 explains the computation and application of the equalizers at a lower level. A simple block Diagram is shown in Figure 4.1.

This chapter will proceed as follows, section 4.2 will explain the algorithm used to find the preambles and packetize the received signal, section 4.3 will explain the frequency offset estimator and frequency offset compensation, section 4.4 will explain channel channel estimation, section 4.5 will explain noise variance, section 4.6 will explain the GPU implementation of the OQPSK detector. The explanation of the GPU implantation of the equalizers will be explained in much detail in Chapter 5.

4.2 Preamble Detection

The received samples in this project has the iNET packet structure shown in Figure 4.2. The iNET packet consists of a preamble and ASM periodically inserted into the data stream. The iNET preamble and ASM bits are inserted every 6144 data bits. The received signal is sampled at 2 samples/bit, making a iNET packet L_{pkt} long or 12672 samples. The iNET preamble comprises eight repetitions of the 16-bit sequence CD98_{hex} and the ASM field

$$034776C72728950B0_{\text{hex}} \quad (4.1)$$

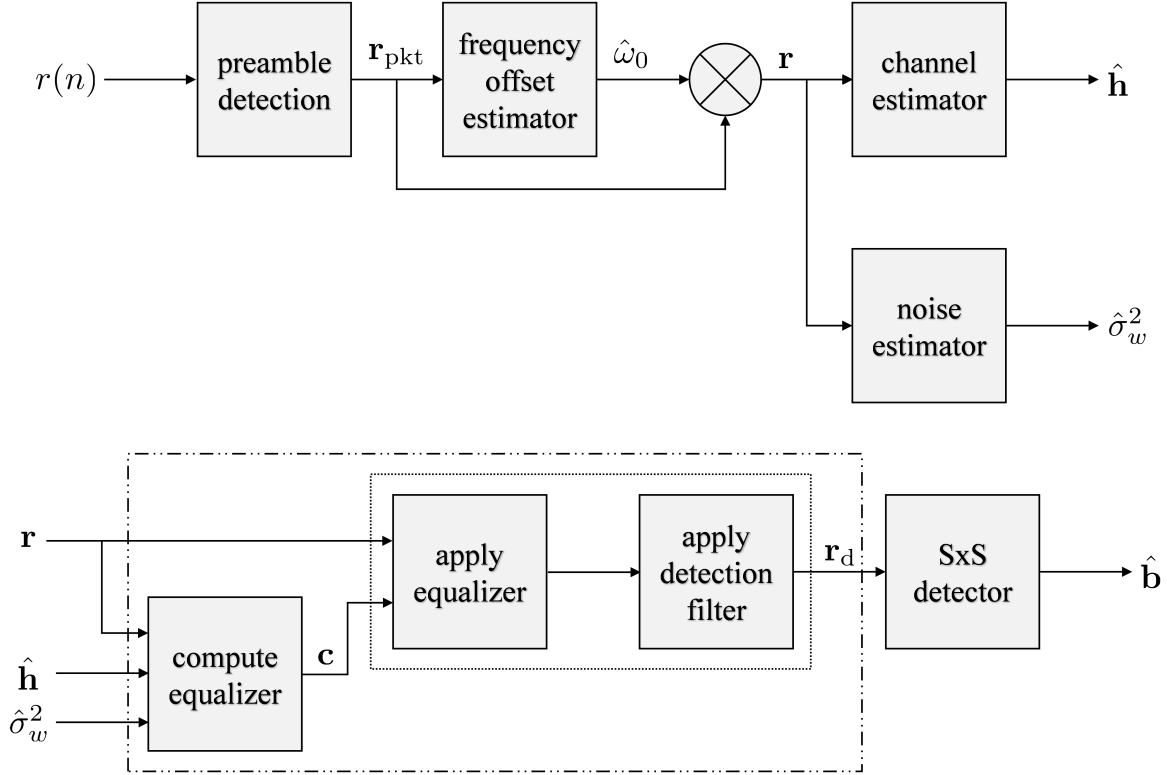


Figure 4.1: This a simple block diagram of what the GPU does.

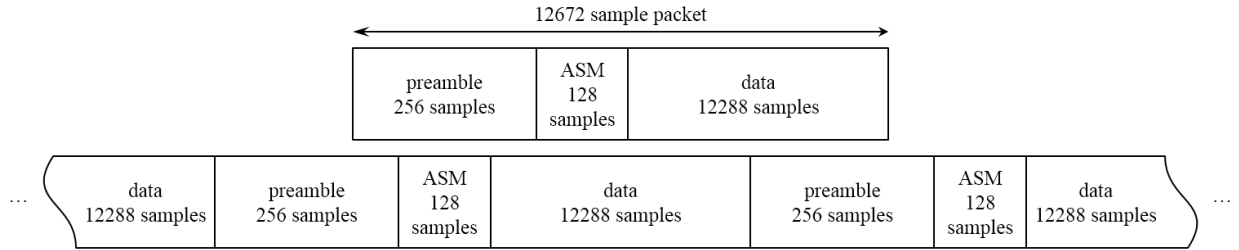


Figure 4.2: The iNET packet structure.

Each 16-bit sequence $CD98_{\text{hex}}$ sampled at two samples/bit are 32 or L_q samples long.

To compute data-aided preamble assisted equalizers, preambles in the received signal are found then used to estimate various parameters. The goal of the preamble detection step is to "packetize" the received samples into vectors with the packet structure shown in Figure 4.2. Each packet of received samples contains a L_p preamble samples, L_{ASM} ASM samples and L_d data

samples. The received signal is sampled at two samples per bit making $L_p = 256$, $L_{ASM} = 136$ and $L_d = 12288$. The full length of a packet is $L_p + L_{ASM} + L_d = 12672$.

Before the received samples can be packetized, the preambles are found using a preamble detector explained in [4]. The preamble detector output $L(u)$ is computed by

$$L(u) = \sum_{m=0}^7 [I^2(n, m) + Q^2(n, m)] \quad (4.2)$$

where the inner summations are

$$\begin{aligned} I(n, m) \approx & \sum_{\ell \in \mathcal{L}_1} r_R(\ell + 32m + n) - \sum_{\ell \in \mathcal{L}_2} r_R(\ell + 32m + n) + \sum_{\ell \in \mathcal{L}_3} r_I(\ell + 32m + n) - \sum_{\ell \in \mathcal{L}_4} r_I(\ell + 32m + n) \\ & + 0.7071 \left[\sum_{\ell \in \mathcal{L}_5} r_R(\ell + 32m + n) - \sum_{\ell \in \mathcal{L}_6} r_R(\ell + 32m + n) \right. \\ & \left. + \sum_{\ell \in \mathcal{L}_7} r_I(\ell + 32m + n) - \sum_{\ell \in \mathcal{L}_8} r_I(\ell + 32m + n) \right], \quad (4.3) \end{aligned}$$

and

$$\begin{aligned} Q(n, m) \approx & \sum_{\ell \in \mathcal{L}_1} r_I(\ell + 32m + n) - \sum_{\ell \in \mathcal{L}_2} r_I(\ell + 32m + n) \\ & - \sum_{\ell \in \mathcal{L}_3} r_R(\ell + 32m + n) + \sum_{\ell \in \mathcal{L}_4} r_R(\ell + 32m + n) \\ & + 0.7071 \left[\sum_{\ell \in \mathcal{L}_5} r_I(\ell + 32m + n) - \sum_{\ell \in \mathcal{L}_6} r_I(\ell + 32m + n) \right. \\ & \left. - \sum_{\ell \in \mathcal{L}_7} r_R(\ell + 32m + n) + \sum_{\ell \in \mathcal{L}_8} r_R(\ell + 32m + n) \right] \quad (4.4) \end{aligned}$$

with

$$\begin{aligned}
\mathcal{L}_1 &= \{0, 8, 16, 24\} \\
\mathcal{L}_2 &= \{4, 20\} \\
\mathcal{L}_3 &= \{2, 10, 14, 22\} \\
\mathcal{L}_4 &= \{6, 18, 26, 30\} \\
\mathcal{L}_5 &= \{1, 7, 9, 15, 17, 23, 25, 31\} \\
\mathcal{L}_6 &= \{3, 5, 11, 12, 13, 19, 21, 27, 28, 29\} \\
\mathcal{L}_7 &= \{1, 3, 9, 11, 12, 13, 15, 21, 23\} \\
\mathcal{L}_8 &= \{5, 7, 17, 19, 25, 27, 28, 29, 31\}.
\end{aligned} \tag{4.5}$$

Figure 4.3 shows $2L_{\text{pkt}}$ samples of the preamble detector output $L(u)$. The start of a preamble is indicated by a local maximum of the preamble detector output. Using the index of the local maximums, the received samples are packetized. The vector \mathbf{r}_{pkt} as shown in Figure 4.1 contains 12672 samples of data with the packet structure shown in Figure 4.2.

The preamble detection algorithm in Equations (4.2)-(4.5) and the local maximum search algorithms are easily implemented into GPUs. The GPU implementation of these algorithms wont be explained here.

4.3 Frequency Offset Compensation

The frequency offset estimator shown in Figure 4.1 is an algorithm taken from blah. With the notation adjusted slightly, the frequency offset estimate is

$$\hat{\omega}_0 = \frac{1}{L_q} \arg \left\{ \sum_{n=i+2L_q}^{i+7L_q-1} r(n)r^*(n-L_q) \right\} \tag{4.6}$$

where L_q is the length of where a frequency offset estimate is produced for every packet in \mathbf{r}_{pkt} .

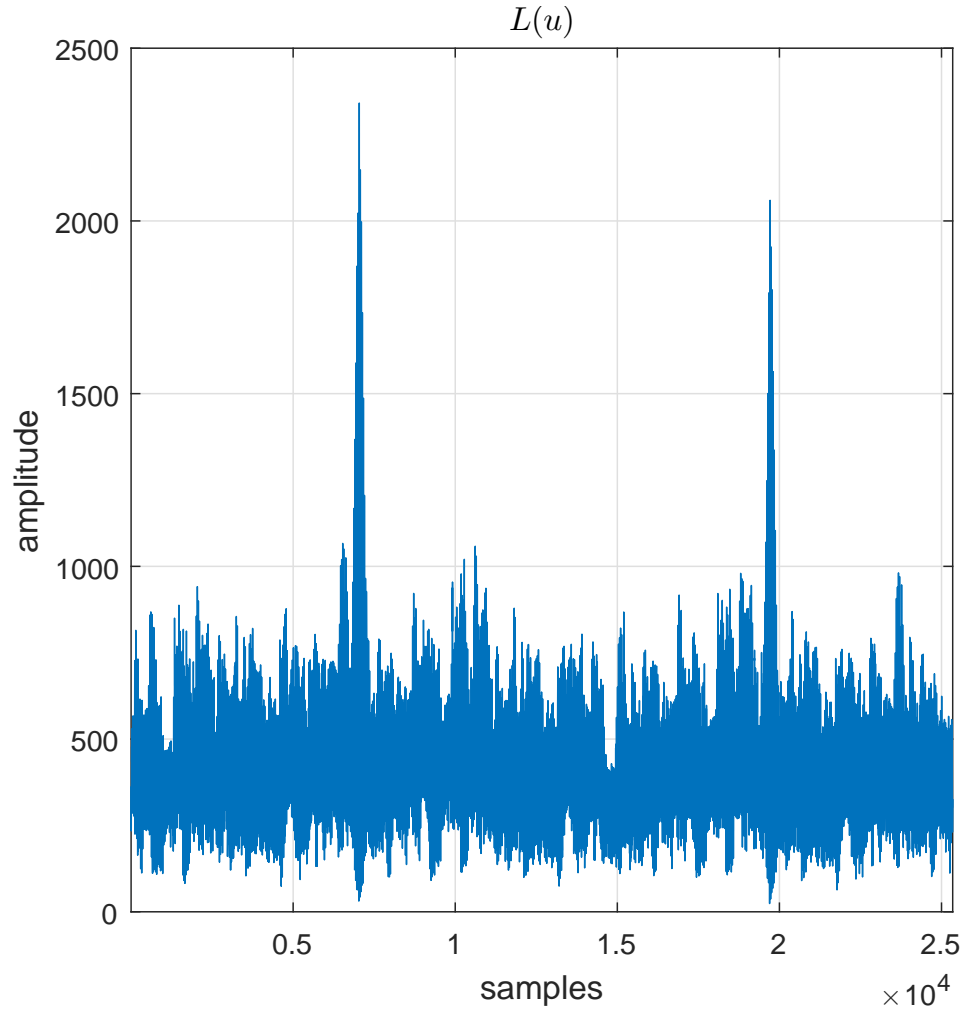


Figure 4.3: The output of the Preamble Detector $L(u)$.

The frequency offset is compensated for by derotating the packetized samples by $-\hat{\omega}_0$

$$r(n) = r_{\text{pkt}}(n)e^{-j\hat{\omega}_0} \quad (4.7)$$

Equations (4.6) and (4.7) are easily implemented into GPUs.

4.4 Channel Estimation

The channel estimator is the ML estimator taken from blah.

$$\hat{\mathbf{h}} = \underbrace{(\mathbf{X}^\dagger \mathbf{X})^{-1}}_{\mathbf{P}_{ix}} \mathbf{X}^\dagger \mathbf{r} \quad (4.8)$$

where \mathbf{X} is a convolution matrix formed from the ideal preamble and ASM samples. The matrix \mathbf{P}_{ix} is

$$\mathbf{P}_{ix} = (\mathbf{X}^\dagger \mathbf{X})^{-1} \mathbf{X}^\dagger \quad (4.9)$$

making the channel estimate simply

$$\hat{\mathbf{h}} = \mathbf{P}_{ix} \mathbf{r} \quad (4.10)$$

The matrix multiplication is easily implemented into GPUs.

4.5 Noise Variance Estimation

The noise variance estimator is the algorithm taken from blah. The algorithm is

$$\hat{\sigma}_w^2 = \frac{1}{2\rho} \left| \mathbf{r} - \mathbf{X} \hat{\mathbf{h}} \right|^2 \quad (4.11)$$

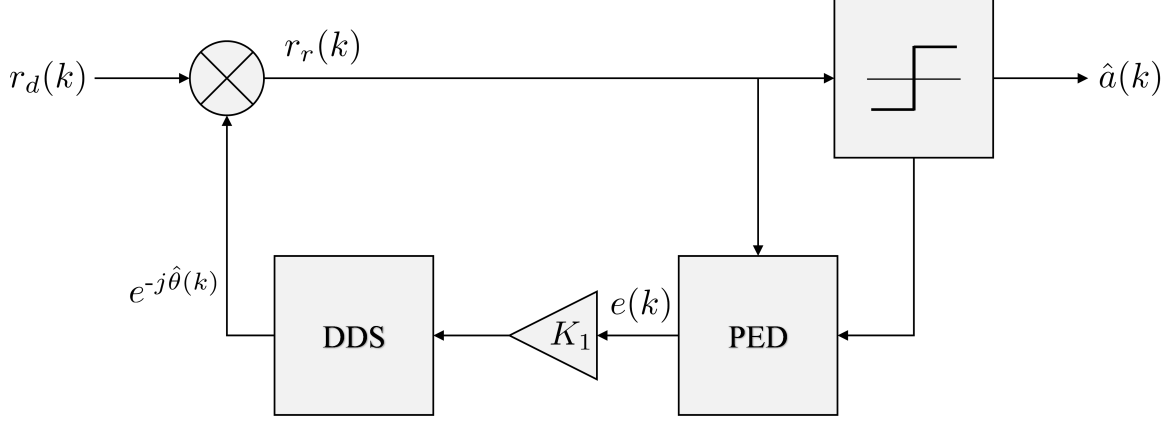
where ρ is the pre-computed constant

$$\rho = \text{Trace} \left\{ \mathbf{I} - \mathbf{X} (\mathbf{X}^\dagger \mathbf{X})^{-1} \mathbf{X}^\dagger \right\}. \quad (4.12)$$

Equation (4.11) is easily implemented into GPUs.

4.6 SxS Detector

The Symbol by Symbol (SxS) detector block in Figure 4.1 is a Offset Quadrature Phase Shift Keying (OQPSK) detector. Using the simple OQPSK detector in place of the complex MLSE SOQPSK-TG detector leads to less than 1 dB in bit error rate blah.



$$\hat{a}(k) = \begin{cases} p(k) & k < L_p + L_{asm} \\ \text{sgn}(\Re\{r_r(k)\}) & k \geq L_p + L_{asm} \quad \& \quad k \text{ even} \\ \text{sgn}(\Im\{r_r(k)\}) & k \geq L_p + L_{asm} \quad \& \quad k \text{ odd} \end{cases}$$

$$e(k) = \begin{cases} 0 & k \text{ even} \\ \hat{a}(k-1)\Im\{r_r(k-1)\} - \hat{a}(k)\Re\{r_r(k)\} & k \text{ odd} \end{cases}$$

Figure 4.4: Offset Quadrature Phase Shift Keying symbol by symbol detector.

The Phase Lock Loop (PLL) in the SxS OQPSK detector cannot be parallelized to be implemented into GPUs because of the feedback loop. Feedback loops are inherently serial. Although the OQPSK detector cannot be parallelized on a sample by sample basis, it can be parallelized on a packet by packet basis. Running the PLL and detector serially through a full packet of data is still relatively fast because each iteration of the PLL and detector is computationally light.

Chapter 5

Equalizer Equations

5.1 Overview

There are 3 different kinds of equalizers I run 1. the solving ones!!! They are equations like $Ax=b$ where I have A and b but I need x 2. the initialized then iterative ones. CMA is initialized with MMSE then runs as many times as possible 3. the multiply ones! the FDEs are a simple multiply in the frequency domain

5.2 Zero-Forcing and MMSE Equalizers

The Zero-Forcing (ZF) and MMSE equalizers are treated together here because they have many common features...

The ZF equalizer is an FIR filter defined by the coefficients

$$c_{ZF}(-L_1) \cdots c_{ZF}(0) \cdots c_{ZF}(L_2). \quad (5.1)$$

The filter coefficients are the solution to the matrix vector equation [?, eq. (324)]

$$\mathbf{H}\mathbf{c}_{ZF} = \mathbf{u}_{n_0} \quad (5.2)$$

where

$$\mathbf{c}_{\text{ZF}} = \begin{bmatrix} c_{\text{ZF}}(-L_1) \\ \vdots \\ c_{\text{ZF}}(0) \\ \vdots \\ c_{\text{ZF}}(L_2) \end{bmatrix}, \quad (5.3)$$

$$\mathbf{u}_{n_0} = \left. \begin{bmatrix} 0 \\ \vdots \\ 0 \\ 1 \\ 0 \\ \vdots \\ 0 \end{bmatrix} \right\} \begin{matrix} n_0 - 1 \text{ zeros} \\ \\ N_1 + N_2 + L_1 + L_2 - n_0 + 1 \text{ zeros} \end{matrix}, \quad (5.4)$$

and

$$\mathbf{H} = \begin{bmatrix} h(-N_1) & & & \\ h(-N_1 + 1) & h(-N_1) & & \\ \vdots & \vdots & \ddots & \\ h(N_2) & h(N_2 - 1) & & h(-N_1) \\ & h(N_2) & & h(-N_1 + 1) \\ & & & \vdots \\ & & & h(N_2) \end{bmatrix}. \quad (5.5)$$

Jeff explains how cuda solvers handle this equation.

The ZF equalizer was studied in the PAQ Phase 1 Final Report in equation 324

$$\mathbf{c}_{\text{ZF}} = (\mathbf{H}^\dagger \mathbf{H})^{-1} \mathbf{H}^\dagger \mathbf{u}_{n_0} \quad (5.6)$$

where \mathbf{c}_{ZF} is a $L_{eq} \times 1$ vector of equalizer coefficients computed to invert the channel estimate \mathbf{h} and \mathbf{u}_{n_0} is the desired channel impulse response centered on $n_0 = N_1 + L_1 + 1$

$$\mathbf{u}_{n_0} = \begin{bmatrix} 0 \\ \vdots \\ 0 \\ 1 \\ 0 \\ \vdots \\ 0 \end{bmatrix} \left. \begin{array}{l} \text{ } \\ \text{ } \\ \text{ } \end{array} \right\} \begin{array}{l} n_0 - 1 \text{ zeros} \\ \\ N_1 + N_2 + L_1 + L_2 - n_0 + 1 \text{ zeros} \end{array} . \quad (5.7)$$

The $L_{eq} + N_1 + N_2 \times L_{eq}$ convolution matrix \mathbf{H} is built using the channel estimate \mathbf{h}

$$\mathbf{H} = \begin{bmatrix} h(-N_1) & & & \\ h(-N_1 + 1) & h(-N_1) & & \\ \vdots & \vdots & \ddots & \\ h(N_2) & h(N_2 - 1) & h(-N_1) & \\ & h(N_2) & h(-N_1 + 1) & \\ & & \vdots & \\ & & & h(N_2) \end{bmatrix} . \quad (5.8)$$

The computation of the coefficients in Equation (5.6) can be simplified in a couple of ways: First the matrix multiplication of \mathbf{H}^\dagger and \mathbf{H} is the autocorrelation matrix of the channel

$$\mathbf{R}_h = \mathbf{H}^\dagger \mathbf{H} = \begin{bmatrix} r_h(0) & r_h^*(1) & \cdots & r_h^*(L_{eq} - 1) \\ r_h(1) & r_h(0) & \cdots & r_h^*(L_{eq} - 2) \\ \vdots & \vdots & \ddots & \\ r_h(L_{eq} - 1) & r_h(L_{eq} - 2) & \cdots & r_h(0) \end{bmatrix} \quad (5.9)$$

where

$$r_h(k) = \sum_{n=-N_1}^{N_2} h(n)h^*(n-k). \quad (5.10)$$

Second the matrix vector multiplication of \mathbf{H}^\dagger and \mathbf{u}_{n_0} is simply the n_0 th row of \mathbf{H}^\dagger or the conjugated n_0 th column of \mathbf{H} . A new vector \mathbf{h}_{n_0} is defined by

$$\mathbf{h}_{n_0} = \mathbf{H}^\dagger \mathbf{u}_{n_0} = \begin{bmatrix} h(L_1) \\ \vdots \\ h(0) \\ \vdots \\ h(-L_2) \end{bmatrix}. \quad (5.11)$$

To simplify, Equations (5.9) and (5.11) are substituted into Equation (5.6) resulting in

$$\mathbf{c}_{ZF} = \mathbf{R}_h^{-1} \mathbf{h}_{n_0}. \quad (5.12)$$

Computing the inverse of \mathbf{R}_h is computationally heavy because an inverse is an N^3 operation. To avoid an inverse, \mathbf{R}_h is moved to the left side and \mathbf{c}_{ZF} is found by solving a system of linear equations. Note that $r_h(k)$ only has support on $-L_{ch} \leq k \leq L_{ch}$ making \mathbf{R}_h sparse or 63% zeros. The sparseness of \mathbf{R}_h is leveraged to reduce computation drastically. The Zero-Forcing Equalizer coefficients are computed by solving

$$\mathbf{R}_h \mathbf{c}_{ZF} = \mathbf{h}_{n_0}. \quad (5.13)$$

MMSE Equalizer

The MMSE equalizer has the same form as the Zero-Forcing equalizer. The MMSE equalizer was also studied in the PAQ Phase 1 Final Report in equation 330.

$$\mathbf{c}_{MMSE} = [\mathbf{G}\mathbf{G}^\dagger + \frac{\sigma_w^2}{\sigma_s^2} \mathbf{I}_{L_1+L_2+1}] \mathbf{g}^\dagger \quad (5.14)$$

where

$$\mathbf{G} = \begin{bmatrix} h(N_2) & \cdots & h(-N_1) \\ & h(N_2) & \cdots & h(-N_1) \\ & & \ddots & \ddots \\ & & & h(N_2) & \cdots & h(-N_1) \end{bmatrix} \quad (5.15)$$

and

$$\mathbf{g} = \begin{bmatrix} h(L_1) \cdots h(-L_2) \end{bmatrix}. \quad (5.16)$$

The vector \mathbf{g}^\dagger is also the same vector as \mathbf{h}_{n_0} in Equation (5.7). The matrix multiplication $\mathbf{G}\mathbf{G}^\dagger$ is also the same autocorrelation matrix \mathbf{R}_h as Equation (5.9). The fraction $\frac{1}{2\sigma_w^2}$ is substituted in for the fraction $\frac{\sigma_w^2}{\sigma_s^2}$ using Equation 333 Rice's report. MMSE only differs from Zero-Forcing by adding the signal-to-noise ratio estimate down the diagonal of the autocorrelation matrix \mathbf{R}_h . Substituting in all these similarities in to Equation (5.14) results in

$$\left[\mathbf{R}_h + \frac{1}{2\sigma_w^2} \mathbf{I}_{L_1+L_2+1} \right] \mathbf{c}_{\text{MMSE}} = \mathbf{h}_{n_0}. \quad (5.17)$$

To further simplify the notation, \mathbf{R}_{hw} is substituted in for $\mathbf{R}_h + \frac{1}{2\sigma_w^2} \mathbf{I}_{L_1+L_2+1}$ where

$$\mathbf{R}_{hw} = \mathbf{R}_h + \frac{1}{2\sigma_w^2} \mathbf{I}_{L_1+L_2+1} = \begin{bmatrix} r_h(0) + \frac{1}{2\sigma_w^2} & r_h^*(1) & \cdots & r_h^*(L_{eq} - 1) \\ r_h(1) & r_h(0) + \frac{1}{2\sigma_w^2} & \cdots & r_h^*(L_{eq} - 2) \\ \vdots & \vdots & \ddots & \\ r_h(L_{eq} - 1) & r_h(L_{eq} - 2) & \cdots & r_h(0) + \frac{1}{2\sigma_w^2} \end{bmatrix}. \quad (5.18)$$

The MMSE equalizer coefficients are solved for in a similar fashion to the Zero-Forcing equalizer coefficients in Equation (5.13).

$$\mathbf{R}_{hw} \mathbf{c}_{\text{MMSE}} = \mathbf{h}_{n_0}. \quad (5.19)$$

5.2.1 The Iterative Equalizer

The Constant Modulus Algorithm

CMA uses a steepest decent algorithm.

$$\mathbf{c}_{b+1} = \mathbf{c}_b - \mu \nabla \mathbf{J} \quad (5.20)$$

The vector \mathbf{J} is the cost function and ∇J is the cost function gradient defined in the PAQ report 352 by

$$\nabla J = \frac{2}{L_{pkt}} \sum_{n=0}^{L_{pkt}-1} \left[y(n)y^*(n) - R_2 \right] y(n) \mathbf{r}^*(n). \quad (5.21)$$

where

$$\mathbf{r}(n) = \begin{bmatrix} r(n + L_1) \\ \vdots \\ r(n) \\ \vdots \\ r(n - L_2) \end{bmatrix}. \quad (5.22)$$

This means ∇J is of the form

$$\nabla J = \begin{bmatrix} \nabla J(-L_1) \\ \vdots \\ \nabla J(0) \\ \vdots \\ \nabla J(L_2) \end{bmatrix}. \quad (5.23)$$

To leverage the computational efficiency of the Fast Fourier Transform (FFT), Equation (5.21) is re-expressed as a convolution.

To begin messaging ∇J

$$z(n) = 2 \left[y(n)y^*(n) - R_2 \right] y(n) \quad (5.24)$$

is defined to make the expression of ∇J to be

$$\nabla J = \frac{1}{L_{pkt}} \sum_{n=0}^{L_{pkt}-1} z(n) \mathbf{r}^*(n). \quad (5.25)$$

then writing the summation out in vector form

$$\nabla J = \frac{z(0)}{L_{pkt}} \begin{bmatrix} r^*(L_1) \\ \vdots \\ r^*(0) \\ \vdots \\ r^*(L_2) \end{bmatrix} + \frac{z(1)}{L_{pkt}} \begin{bmatrix} r^*(1+L_1) \\ \vdots \\ r^*(1) \\ \vdots \\ r^*(1-L_2) \end{bmatrix} + \dots + \frac{z(L_{pkt}-1)}{L_{pkt}} \begin{bmatrix} r^*(L_{pkt}-1+L_1) \\ \vdots \\ r^*(L_{pkt}-1) \\ \vdots \\ r^*(L_{pkt}-1-L_2) \end{bmatrix}. \quad (5.26)$$

The k th value of ∇J is

$$\nabla J(k) = \frac{1}{L_{pkt}} \sum_{m=0}^{L_{pkt}-1} z(m) r^*(m-k), \quad -L_1 \leq k \leq L_2. \quad (5.27)$$

The summation almost looks like a convolution. To put the summation in convolution form, define

$$\rho(n) = r^*(n). \quad (5.28)$$

Now

$$\nabla J(k) = \frac{1}{L_{pkt}} \sum_{m=0}^{L_{pkt}-1} z(m) \rho(k-m). \quad (5.29)$$

Because $z(n)$ has support on $0 \leq n \leq L_{pkt} - 1$ and $\rho(n)$ has support on $-L_{pkt} + 1 \leq n \leq 0$, the result of the convolution sum $b(n)$ has support on $-L_{pkt} + 1 \leq n \leq L_{pkt} - 1$. Putting all the pieces together, we have

$$b(n) = \sum_{m=0}^{L_{pkt}-1} z(m) \rho(n-m)$$

$$= \sum_{m=0}^{L_{pkt}-1} z(m)r^*(m-n) \quad (5.30)$$

Comparing Equation (5.29) and (5.30) shows that

$$\nabla J(k) = \frac{1}{L_{pkt}} b(k), \quad -L_1 \leq k \leq L_2. \quad (5.31)$$

The values of interest are shown in Figure Foo!!!!(c)

This suggest the following algorithm for computing the gradient vector ∇J Matlab Code!!!

5.2.2 The Multiply Equalizers

The Frequency Domain Equalizer One

The Frequency Domain Equalizer One (FDE1) is the MMSE or wiener filter applied in the frequency domain. Ian E. Williams and M. Saquib derived FDE1 for this project in a paper called Linear Frequency Domain Equalization of SOQPSK-TG for Wideband Aeronautical Telemetry. The FDE1 equalizer is defined in Equation (11) as

$$C_{\text{FDE1}}(\omega) = \frac{\hat{H}^*(\omega)}{|\hat{H}(\omega)|^2 + \frac{1}{\hat{\sigma}^2}} \quad (5.32)$$

The term $C_{\text{FDE1}}(\omega)$ is the Frequency Domain Equalizer One frequency response at ω . The term $\hat{H}(\omega)$ is the channel estimate frequency response at ω . The term $\hat{\sigma}^2$ is the noise variance estimate, this term is completely independent of frequency because the noise is assumed to be white or spectrally flat.

FDE1 needs no massaging because Equation (5.32) is easily implemented in the GPU and it is computationally efficient.

The Frequency Domain Equalizer One

The Frequency Domain Equalizer Two (FDE2) is the MMSE or wiener filter applied in the frequency domain. Ian E. Williams and M. Saquib derived FDE1 for this project in a paper called Linear Frequency Domain Equalization of SOQPSK-TG for Wideband Aeronautical Telemetry. The FDE2 equalizer is defined in Equation (12) as

$$C_{\text{FDE2}}(\omega) = \frac{\hat{H}^*(\omega)}{|\hat{H}(\omega)|^2 + \frac{\Psi(\omega)}{\hat{\sigma}^2}} \quad (5.33)$$

FDE2 almost identical to FDE1. The only difference is term $\Psi(\omega)$ in the denominator. The term $\Psi(\omega)$ is the average spectrum of SPQOSK-TG shown in Figure 5.1. FDE2 needs no massaging because Equation (5.33) is easily implemented in the GPU and is computationally efficient.

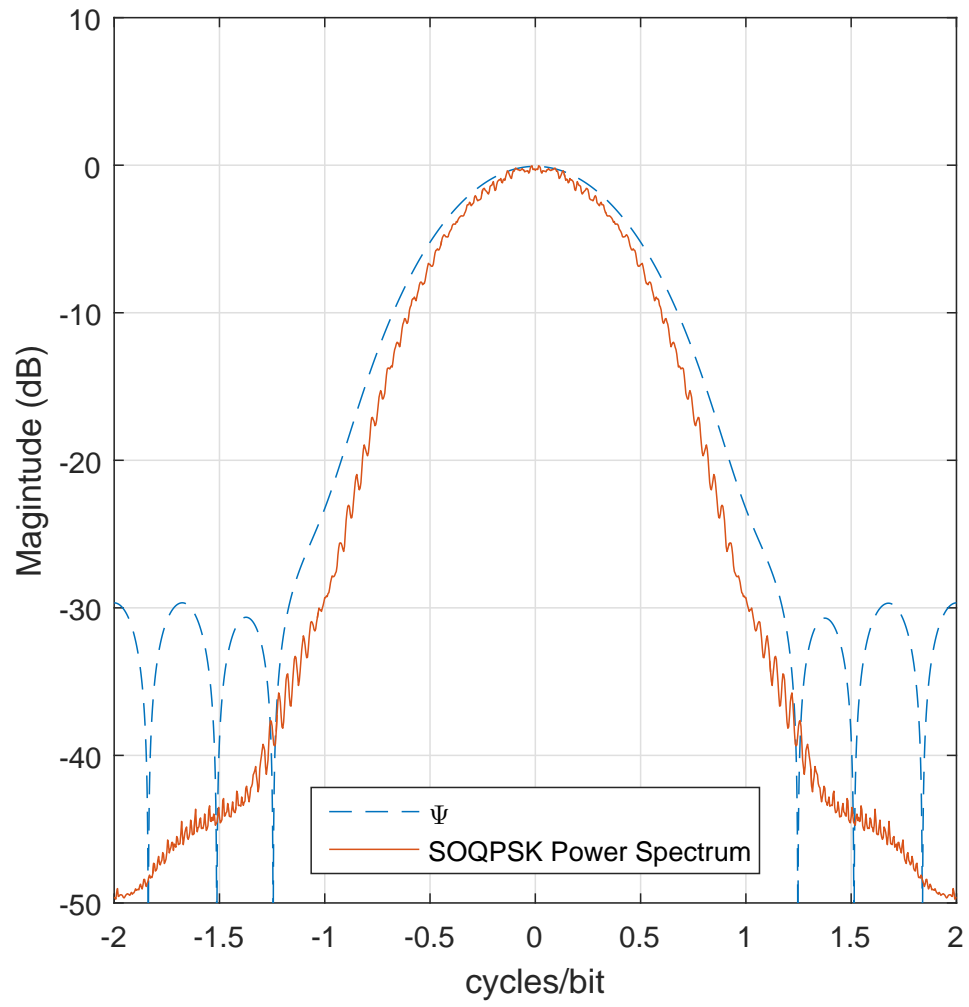


Figure 5.1: A block diagram illustrating organization of the algorithms in the GPU.

Chapter 6

Equalizer Performance

This is the Equalizer Performance

Chapter 7

Final Summary

this is the final summary

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