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ABSTRACT

Nowadays, always on, high speed internet connections are becoming popular due to technologies like DSL and Cable making network security a critical factor for the success of many applications. Network Intrusion Detection Systems (NIDS) are based on pattern matching techniques applied to the incoming packets. These systems can check both the header and the body of the packet for better results in detecting security threats. Of course, checking the body of the packet against known attacks requires great deal of processing power and if it's not done fast enough it can introduce a bottleneck to the system's performance.

NIDS can be implemented in hardware or software. Both ways have advantages and disadvantages. Hardware based solutions use ASICs or FPGAs. They generally outperform software systems in terms of pattern matching speed. FPGAs are more flexible than ASICs since it's easier to be reprogrammed and thus allowing updates of the rule set, while ASICs use integrated processors with large memories allowing the development of more complex code. On the other hand, software systems offer even greater flexibility since they can be extended in any possible way, in order to efficiently face new kinds of attacks, such as rule set update and addition of new pattern matching techniques.

This diploma thesis studies the use of hardware NIDS. Based on the work of Sourdis & Pnevmatikatos on pre-decoded CAMs, we explore the use of minimum cut (min-cut) algorithms to further increase the speed and reduce the complexity of pattern matching in the body of the packet.

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Chapter 1

1.1 Introduction

Intrusion detection systems are an important component of defensive measures protecting computer systems and networks from abuse. Although intrusion detection technology is still and should not be considered as a complete defence, we believe it can play a significant role in an overall security architecture. If an organization chooses to deploy an IDS, a range of commercial and public domain products are available that offer varying deployment costs and potential to be effective. When an IDS is properly deployed, it can provide warnings indicating that a system is under attack, even if the system is not vulnerable to the specific attack. These warnings can help users alter their installation's defensive posture to increase resistance to attack. In addition, an IDS can serve to confirm secure configuration and operation of other security mechanisms such as firewalls.

1.1.1 Scope of this thesis

Network Intrusion Detection Systems (NIDS) perform deep packet inspection. They scan packet's payload looking for patterns that would indicate security threats. Matching every incoming byte, though, against thousands of pattern characters at wire rates is a complicated task. So, string matching can be considered as one of the most computationally intensive parts of a NIDS. Many different algorithms or combination of algorithms have been introduced and implemented in general purpose processors (GPP) for fast string matching using as datasets rulesets from the SNORT NIDS [29].

This thesis is based on the work of Sourdis and Pnevmatikatos ([19], [20]) where they exploited the fact that FPGAs are flexible, reconfigurable devices, fast enough for implementing such systems. One of the main drawbacks in FPGA's is that the matching of a large number of patterns has high area cost, so sharing logic is critical, since it could save a significant amount of resources, and make designs smaller and faster.

Since string matching is the most computationally intensive part of an NIDS, our proposed solution maintains high performance and minimizes area cost. Partitioning the entire ruleset of search patterns in smaller groups, we can implement the entire match logic for each of these groups in a much smaller area reducing the average length of the wires.

1.1.2 Outline of this thesis

The rest of this thesis is organized as follows: the rest of this chapter gives a thorough description of Network Intrusion Detection Systems. In Chapter 2 we present the background work of a hardware-based NIDS developed by Sourdis and Pnevmatikatos, presenting also its performance and cost and discuss its advantages and disadvantages. Next we introduce graph partitioning algorithms in detail and present our solution in the partitioning of the SNORT rule set, aiming at the sharing of logic for each of these groups for reducing the area cost in the implementation. Finally we present the conclusions of this work and discuss future extensions.

1.2 Intrusions and Detection

As e-commerce sites become attractive targets and the emphasis turns from break-ins to **denials of service (DoS)**, the situation will likely worsen. Many early attackers simply wanted to prove that they could break into systems; increasingly nowadays, the trend is toward intrusions motivated by financial, political, and military objectives. In the 1980s, most intruders were experts, with high levels of expertise and individually developed methods for breaking into systems. They rarely used automated tools and exploit scripts. Today, anyone can attack Internet sites using readily available intrusion tools and exploit scripts that capitalize on widely known vulnerabilities. **Figure 1**, taken from Washington Post, which describes the attacks, illustrates the relationship between the relative sophistications of attacks and attackers from the 1980s to the present.



Figure 1.1.: Attack sophistication versus intruder technical knowledge.

Today, damaging intrusions can occur in a matter of seconds. Intruders hide their presence by installing modified versions of system monitoring and administration commands and by erasing their tracks in audit and log files. In the 1980s and early 1990s, denial-of-service (DoS) attacks were infrequent and not considered serious. Today, successful denial-of-service attacks can put e-commerce-based organizations such as online stockbrokers and retail sites out of business. Successful IDSs can recognize both intrusions and denial-of-service activities and invoke countermeasures against them in real time. To realize this potential, we'll need more accurate detection and reduced falsealarm rates.

1.3 What is Intrusion Detection?

Intrusion Detection is a set of techniques and methods that are used to detect suspicious activity both at the network and host level. Intrusion detection systems fall into two basic categories: **signature-based** intrusion detection systems and **anomaly** **detection** systems. Intruders are recognized by signatures like computer viruses that can be detected using software. You try to find data packets that contain any known intrusion related signatures or anomalies related to internet protocols.

Based upon a set of signatures and rules, the detection systems are able to find and log suspicious activity and generate alerts. Anomaly based detection systems usually depends on packet anomalies present in protocol header parts. In some cases these methods produce better results compares to signature based IDS. Usually an intrusion detection system captures data from the network and applies its rules to that data or detects anomalies in it. Snort is primarily rule based IDS, however input plug-ins are present to detect anomalies in protocol headers.



Figure 1-2: Block diagram of a complete network intrusion detection system consisting of Snort, MySQL, Apache, ACID, PHP, PHPLOT, GD Library.

Snort uses rules in text files that can be modified by a text editor. Rules are grouped into categories. Rules that belong to **each category** are stored in **separate files**. These files are then included into a main configuration file called snort.conf. Snort reads

these rules at the start-up time and builds internal data structures or chains to apply these rules to captured data. Finding signatures and using them as rules is a tricky job, since the more rules you use, the more processing power is required to process captured data at real time. It is important to implement as many signatures as you can using as few rules as possible. Snort comes with a rich set of predefined rules to detect intrusion activity and the user is free to add its own rules at will. Some of the built-in rules can also be removed to avoid false alarms.

1.4 Why Use Intrusion Detection?

Intrusion detection devices are an integral part of any network. The Internet is constantly evolving, and new vulnerabilities and exploits are found regularly. Network monitoring tools, worms and viruses, scripts and more are constantly probing the machines and the network. Intrusion systems provide an additional level of protection to detect the presence of an intruder, and help to **provide accountability for the attacker's actions**.

Intrusion detection allows organizations to protect their systems from the threats that come with increasing network connectivity and reliance on information systems. Given the level and nature of modern network security threats, the question for security professionals should not be whether to use intrusion detection, but which intrusion detection features and capabilities to use.

IDSs have gained acceptance as a necessary addition to every organization's security infrastructure. Despite the documented contributions intrusion detection technologies make to system security, in many organizations one must still justify the acquisition of IDSs. There are several compelling reasons to acquire and use IDSs:

1. To prevent problem behaviours by increasing the perceived risk of discovery and punishment for those who would attack or otherwise abuse the system 2. To detect attacks and other security violations that are not prevented by other security measures

3. To detect and deal with the preambles to attacks (commonly experienced as network probes and other "doorknob rattling" activities)

4. To document the existing threat to an organization

5. To act as quality control for security design and administration, especially of large and complex enterprises

6. To provide useful information about intrusions that do take place, allowing improved diagnosis, recovery, and correction of causative factors.

1.4.1. Preventing problems by increasing the perceived risk of discovery and punishment of attackers

A fundamental goal of computer security management is to affect the behaviour of individual users in a way that protects information systems from security problems. Intrusion detection systems help organizations accomplish this goal by increasing the perceived risk of discovery and punishment of attackers. This serves as a significant prohibitive to those who would violate security policy.

1.4.2. Detecting problems that are not prevented by other security measures

Attackers, using widely publicized techniques, can gain unauthorized access to many, if not most systems, especially those connected to public networks. This often happens when known vulnerabilities in the systems are not corrected. Although vendors and administrators are encouraged to address vulnerabilities, (e.g. through public services such as ICAT, http://icat.nist.gov) there are many situations in which this is not possible:

• In many legacy systems, the operating systems cannot be patched or

updated.

• Even in systems in which patches can be applied, administrators sometimes have neither sufficient time nor resource to track and install all the necessary patches. This is a common problem, especially in environments that include a large number of hosts or a wide range of different hardware or software environments.

• Users can have compelling operational requirements for network services and protocols that are known to be vulnerable to attack.

• Both users and administrators make errors in configuring and using systems.

• In configuring system access control mechanisms to reflect an organization's procedural computer use policy, incompatibilities almost always occur. These differences allow legitimate users to perform actions that are ill advised or that overstep their authorization.

1.4.3. Detecting the preambles to attacks (often experienced as network probes and other tests for existing vulnerabilities)

When crackers attack a system, they typically do so in predictable stages.

The first stage of an attack is usually **probing** or **examining** a system or network, searching for an optimal point of entry. In systems with no IDS, the attacker is free to thoroughly examine the system with little risk of discovery or retribution. Given this unfettered access, a determined attacker will eventually find vulnerability in such a network and exploit it to gain entry to various systems.

The same network with an IDS monitoring its operations presents a much more difficult challenge to that attacker. Although the attacker may probe the network for weaknesses, the IDS will observe the probes, will identify them as suspicious, may actively block the attacker's access to the target system, and will alert security personnel who can then take appropriate actions to block subsequent access by the attacker. Even the presence of a reaction to the attacker's probing of the network will elevate the level of risk the attacker perceives, discouraging further attempts to target the network.

1.4.4. Documenting the existing threat

When you are drawing up a budget for network security, it often helps to substantiate claims that the network is likely to be attacked or is even currently under attack. Furthermore, understanding the frequency and characteristics of attacks allows you to understand what security measures are appropriate to protect the network against those attacks. IDSs verify, itemize, and characterize the threat from both outside and inside your organization's network, assisting you in making sound decisions regarding your allocation of computer security resources. Using IDSs in this manner is important, as many people mistakenly deny that anyone (outsider or insider) would be interested in breaking into their networks. Furthermore, the information that IDSs give you regarding the source and nature of attacks allows you to make decisions regarding security strategy driven by demonstrated need, not guesswork.

1.4.5. Quality control for security design and administration

When IDSs run over a period of time, patterns of system usage and detected problems can become apparent. These can highlight flaws in the design and the security for the system, so administrators can correct those deficiencies before they cause an incident.

1.4.6. Providing useful information about actual intrusions

Even when IDSs are not able to block attacks, they can still collect relevant, detailed and trustworthy information about the attack that supports incident handling and recovery efforts. Ultimately, such information can identify problem areas in the organization's security configuration or policy.

1.5 Some definitions

Before we go into details of intrusion detection we need to present some definitions related to security.

1.5.1 IDS

Intrusion Detection System (IDS) is software, hardware or combination of both used to detect intruder activity. Snort is an open source IDS available to the general public. An IDS may have different capabilities depending upon how complex and sophisticated the components are.

1.5.2 NIDS

Network Intrusion detection systems (NIDS) are systems that capture data packets traveling on the network media (cables, wireless) and match them to a database of signatures. Depending on whether a packet is matched with an intrusion signature an alert is generated or the packet is logged into a file or a database. One major use of Snort is as a NIDS. Firewalls, on the other hand, are configured to allow or deny access to a particular service or host based on a set of rules. If the traffic matches an acceptable pattern, it is permitted regardless of what the packet contains. However, a NIDS captures and inspects all traffic regardless of whether it's permitted or not. Based on the contents, at either the **IP** or **application level**, an alert is generated.

One question that arises when deploying NIDSs is where to locate the system "sensors". There are many options for placing a NIDS with different advantages associated with each location:

Location: Behind each external firewall (See Figure 1.4 – Location 1)

Advantages:

- Sees attacks, originating from the outside world, that penetrate the network's perimeter defences.
- Highlights problems with the network firewall policy or performance
- Sees attacks that might target the web server or ftp server
- Even if the incoming attack is not recognized, the IDS can sometimes recognize the outgoing traffic that results from the compromised server



Figure 1.4: Location: Behind each external firewall

Location: Outside an external firewall (See Figure 1.4 – Location 2)

Advantages:

- Documents number of attacks originating on the Internet that target the network.
- Documents types of attacks originating on the Internet that target the network

Location: On major network backbones (See Figure 4 – Location 3)

Advantages:

- Monitors a large amount of a network's traffic, thus increasing the possibility of spotting attacks.
- Detects unauthorized activity by authorized users within the organization's security perimeter.

Location: On critical subnets (See Figure 4 – Location 4)

Advantages:

- Detects attacks targeting critical systems and resources.
- Allows focusing of limited resources to the network assets considered of greatest value.

1.5.3 HIDS

Host-based Intrusion detection systems (HIDS) are installed as agents in a host. These intrusion detection systems can look into system and application files to detect any intruder activity. Some of these systems are reactive, meaning that they inform you only when something has happened. Some HIDS are proactive, they can sniff the network traffic coming to a particular host on which the HIDS is installed and alert you in real time. Some HIDS can also listen to port activity and alert when specific ports are accessed, this allows for some network type attack detection. The HIDS does not require additional hardware to do intrusion detection. Is easily resides on existing network resources (File Servers, Web Servers).

Another consideration when using HIDS is that of allowing operators to become familiar with the IDS in a sheltered, but active environment. Much of the effectiveness of any IDS, but particularly a host-based IDS depends on the operator's ability to distinguish between true and false alarms. Over a period of time, an operator, working with an IDS in a particular environment, will gain a sense of what is normal for that environment, as monitored by the IDS. It is also important (as HIDS are often not continuously attended by operators) to establish a schedule for checking the results of the IDS. If this is not done, the risk that a cracker will tamper with the IDS during an attack increases.

1.5.4 SIDS

Stack-based Intrusion detection systems (SIDS) are the newest technology and vary from vendor to vendor. SIDS works by integrating closely with the TCP/IP stack, allowing packets to be watched as they traverse their way up the OSI layers. Watching the packets in this way allows the IDS to pull the packets from the stack before the OS or the application has a chance to process the packets. To be complete, a SIDS should watch both incoming and outgoing network traffic on a system. By monitoring network packets destined only for a simple host, makes the IDS have sufficiently low overhead so that every system on the network can run SIDS.



Figure 1-5: Multiple Snort sensors in the enterprise, logging to a centralized database server

1.6 Strengths and Limitations of IDSs

Although IDSs are a valuable addition to an organization's security infrastructure, there are things they do well, and other things they do not do well. As an administrator plans the security strategy for his organization's systems, it is important to understand what IDSs should be trusted to do and what goals might be better served by other types of security mechanisms.

1.6.1 Strengths of Intrusion Detection Systems

IDSs perform the following functions well:

- Monitoring and analysis of system events and user behaviours
- Testing the security states of system configurations
- Base-lining the security state of a system, then tracking any changes to that baseline
- Recognizing patterns of system events that correspond to known attacks
- Recognizing patterns of activity that statistically vary from normal activity
- Managing operating system audit and logging mechanisms and the data they generate
- Alerting appropriate staff by appropriate means when attacks are detected.
- Measuring enforcement of security policies encoded in the analysis engine
- Providing default information security policies
- Allowing non-security experts to perform important security monitoring functions.

1.6.2. Limitations of Intrusion Detection Systems

IDSs cannot perform the following functions:

- Compensating for weak or missing security mechanisms in the protection infrastructure. Such mechanisms include firewalls, identification and authentication, link encryption, access control mechanisms, and virus detection and eradication. Instantaneously detecting, reporting, and responding to an attack, when there is a heavy network or processing load.
- Detecting newly published attacks or variants of existing attacks.
- Effectively responding to attacks launched by sophisticated attackers
- Automatically investigating attacks without human intervention.
- Resisting attacks that are intended to defeat or circumvent them
- Compensating for problems with the fidelity of information sources
- Dealing effectively with switched networks.

Chapter 2 Background on Decoded CAM (DCAM) architecture 2.1 Introduction

High speed and always-on network access is becoming commonplace around the world, creating a demand for increased network security. Network Intrusion Detection Systems (NIDS) such as Snort attempt to detect and prevent attacks from the network using pattern-matching rules in a way similar to anti-virus software. These systems must operate at line (wire) speed so that they do not become a bottleneck to the system's performance Network Intrusion Detection Systems running in general purpose processors can only serve up to a few hundred Mbps throughput. Measurements on Snort show that 31% of total processing and 80% in the case of web-intensive traffic is due to string matching. Therefore, string matching can be considered as the most computational intensive part of a NIDS.

Many different algorithms or combinations of algorithms have been introduced and tested on Snort's ruleset but many of these solutions can **only serve up to a few hundred Mbps throughput**. Until now several hardware based solutions have given very promising results, from ASIC commercial products to FPGA-based string matching or finite automata string matching techniques, but not all implementations have achieved high throughput and reasonable area cost like the **FPGA-based implementation in earlier work of Sourdis and Pnevmatikatos ([19], [20])**.

FPGA-based platforms presented so far provide higher flexibility compared to ASIC implementations. FPGA can exploit the fact that the NIDS rules change (relatively infrequently of course) and use reconfiguration to reduce implementation cost. In addition, FPGA-based systems can exploit parallelism in order to achieve satisfactory processing throughput. The use of parallelism (processing multiple bytes or characters per cycle) in general is difficult in finite-automata implementations which are built with the implicit assumption that the input is checked one byte at a time.

In this section we will provide a description of the background work in CAM and DCAM architectures, in order to give a better insight on how our work in this thesis has influenced the performance in the proposed architectures by Sourdis and Pnevmatikatos ([19],[20]). All figures presented in this chapter are borrowed from [19],[20]. We will discuss the basic CAM architecture and how this idea was extended to the DCAM architecture.

2.2 CAM (Content Addressable Memory)

A CAM (content-addressable memory) is a memory device that accelerates any application requiring fast searches of a database, list or pattern, such as in database machines, image or voice recognition, or computer and communication networks. CAMs supply the performance advantage over other memory search algorithms, such as binary or tree-based searches or look-aside tag buffers, by comparing the desired information against the entire list of pre-stored entries simultaneously, giving an order-of-magnitude reduction in the search time.

Thus the term **associative memory** tends to denote forms of association different from familiar ones-forms that presumably have less sharp constraints imposed by the structure of memory (as opposed to the structure of the information in the memory).

In a CAM, data is stored in locations in a somewhat random fashion. The locations can be selected by an address bus or the data can be written directly into the first empty location, because every location has a pair of special status bits that keep track of whether the location has valid information in it or is empty and available for overwriting.

Once information is stored in a memory location, it is found by comparing every bit in memory with data placed in a special comparator register. If there is a match for every bit in a location with every corresponding bit in the comparand, a match flag is asserted to let the user know that the data in the comparand was found in memory. A priority encoder sorts out which matching location has the top priority, if there is more than one, and makes the address of the matching location available to the user. Thus, with a CAM, the user supplies the data and gets back the address. CAMs are based on memory cells that have been modified by the addition of extra transistors that compare the state of the bit stored with the state stored in a comparand register. Logically, CAMs perform an exclusive-NOR function, so that a match is only indicated if both the stored bit and the corresponding comparand bit are the same state.

CAMs can accelerate any application requiring fast searches of databases, lists, or patterns, such as in image or voice recognition, or computer and communication designs. For this reason, **CAMs are used in applications where search time is critical and must be very short.** In each one of these applications the user may not know the addresses of words that have particular pieces of information stored within a specific portion of the word length. For example, the search key could be the IP address of a network user, and the associated information could be a user's access privileges and location on the network. If the search key presented to the CAM is present in the CAM's table, the CAM indicates a match and returns the associated information, which consists of the user's privileges. A CAM can thus operate as a data-parallel or single instruction/multiple data (SIMD) processor.

In [19] Sourdis and Pnevmatikatos have shown that a CAM implemented using discrete comparators for pattern matching has several advantages:

- (i) it is simple and regular
- (ii) it allows for fine grain pipelining and high operating frequencies
- (iii) it is straightforward to use multiple comparators in order to process multiple input bytes per cycle (parallelism)

The main idea behind the CAM comparator structure is that the pattern matching system is organized as a single input that supplies the input stream of characters and an output that is simply an indicator showing if a match occurred, plus an identifier of the matching rule.

The main drawback of this approach is its area cost which is around 4-5 logic cells per search pattern character including all overheads. To reduce the cost they have proposed sharing the result of comparators when the same character was searched for in two different patterns but at the same location. For example in search strings "AB" and "AC" one could use only one comparator for "A" instead of two (Figure 2.1).



Figure 2.1 (taken from paper [20]): Basic CAM comparator structure and optimization. Part (a) shows the straightforward implementation where a shift register holds the last N characters of the input stream. Each character is compared against the desired value (in two nibbles to fit in FPGA LUTs) and all the partial matches are combined with an AND gate to produce the final match result. Part (b) on the other hand illustrates the proposed optimization where the match "A" signals are shared across the two search strings "AB" and "AC" to reduce area cost

In this approach, the input stream is inserted in a shift register and the individual entries are fanout to the pattern comparators. So, to search for strings "AB" and "AC", we have two comparators fed from the first two position of the shift register. Figure 2.1(a) reflects the FPGA implementation where each 8-bit comparator is broken down to two 4-bit comparators each of which fits in one LUT. This implementation is simple and regular and with proper use of pipelining can achieve very high operating frequencies. As we have already mentioned, its drawback is the high area cost. To improve this cost, the solution suggested was **sharing the character comparators for strings with "similarities".** This is shown in Figure 2.1(b) where the result of a single comparator for character A is shared between the two search strings "AB" and "AC".

A complete Intrusion Detection Systems (IDS) based on the Snort rules requires a system optimized for hundreds of rules, many of which require string matching against the entire data segment of a packet. Highly parallel hardware backend technology has been developed in the past, to dramatically increase the speed of string matching, specifically directed toward intrusion detection and response applications. The high level of performance that they provide is necessary to provide real-time string matching at Internet speeds.

Snort has thousands of content-based rules. Each of these rules requires that a packet be searched in its entirety for the occurrence of some "fingerprint" string. Using naive methods, this is unworkable. Using more sophisticated algorithms or higher levels of parallelism, it becomes tenable.

Sourdis and Pnevmatikatos [20] have developed a pattern-matching co-processor that supports all pattern matching functions of the Snort rule language. In order to achieve maximum pattern capacity and throughput, the design focuses on minimizing circuit area while maintaining high clock speed.

2.3 DCAM (pre-Decoded Content Addressable Memory)

The next work by Sourdis and Pnevmatikatos **extends the idea of CAM** further: instead of keeping a window of input characters in the shift register each of which is compared against search patterns, equality of the input for the desired characters can be tested firstly, and then delay the partial matching signals.

These two approaches (sharing the character comparison and delaying partial matching) are compared in Figure 2.2. In this Figure, part (a) corresponds to the earlier design with the LUT details abstracted away in the equality boxes and part (b) showing how the equality is tested primary for the three distinct characters followed by a postponement in the matching of character A to obtain the complete match for strings "AB" and "AC". This approach achieves not only the sharing of the equality logic for character A, but also transforms the 8-bit wide shift register used in part (a) into possibly multiple single bit shift register for the equality result. Hence, if we can exploit this advantage, the potential for area savings is significant. This architecture design is called DCAM.



Figure 2.2 (taken from paper [20]) : Comparator Optimization: starting from the shared comparator implementation of Figure 1 the comparators are placed **before** the shift register, and the matching of signal is delayed to achieve the correct result. Note that the shift register is 8-bit wide in part (a), and 1-bit wide part (b).

One thing that was taken into consideration in this DCAM implementation is that the number of single bit shift registers is proportional to the length of the search patterns. In Figure 3.3 we can see how shift registers affect the architecture design.



Figure 2.3 (taken from paper [20]): To match the string "ABCD" we have to remember the matching of character A 3 cycles ago, the matching of B two cycles ago, etc, until the final character is matched in the current cycle.

To match a string of length four characters, we (i) need to test equality for these four characters (in the dashed "decoder" block), and (ii) to delay the matching of the first character by three cycles, the matching of the second character by two cycles, and so on, for the width of the search pattern. In total, the number of storage elements required in this approach is L * (L - 1)/2 for a string of length L. To overcome this disadvantage, the number of shift registers was reduced by sharing their outputs whenever the same character is used in the same position in multiple search patterns, and secondly an optimized implementation of a shift register was used with a device (Xilinx) that uses a single logic cell for a shift register.

2.4 Practices to increase performance

In order to achieve better performance, two basic techniques were used to improve the operating speed as well as the throughput of the DCAM implementation. To achieve high operating frequency, parallelism was used and **to achieve better performance and area density, partitioning was used**.

We will only discuss the partitioning technique here since this is the purpose of this thesis. The main idea on how the partitioning method influences performance was introduced in [20] and will be discussed in the following section. This thesis focuses on a software implementation (METIS) of a multilevel recursive bisection algorithm to achieve the partitioning of the entire search pattern rule set into smaller groups.

In terms of performance, a limiting factor to the scaling of an implementation to a large number of search patterns is the fanout and the length of the interconnections. For example, if we consider a set of search patterns with 10,000 uniformly distributed characters, we have an average fanout of 40 for each of the decoders' outputs. Furthermore, the distance between all the decoders outputs and the equality checking AND gates will be significant.

If we partition the entire set of search patterns in smaller groups, we can implement the entire fanout-decode-match logic for each of these groups in a much smaller area, reducing the average length of the wires. This reduction in the wire length though comes at the cost of multiple decoders. With grouping, we need to decode a character for each of the group in which they appear, increasing the area cost. On the other hand, the smaller groups may require smaller decoders, if the number of distinct characters in the group is small. Hence, if we group together search patterns with more similarities we can reclaim some of the multi-decoder overhead.

In the partitioned design, each of the partitions will have a structure similar to the one depicted in Figure 2.4. The multiple groups will be fed data through a fanout tree, and all the individual matching results will be combined to produce the final matching output. Each of the partitions will be relatively small, and hence can operate at a high frequency. However, for large designs, the fanout of the input stream much traverse long distances.

In their designs Sourdis and Pnevmatikatos observe that these long wires sometimes limit the frequency for the entire design. To tackle this bottleneck they used multiple clocks: one slow clock to distribute the data across long distances over wide busses, and a fast clock for the smaller and faster partitioned matching function. Experimenting with various partition sizes and slow-to fast clock speed ratios they observed that reasonable sizes for groups is between 64 and 256 search patterns, while a slow clock of twice the period is slow enough for their designs.



Figure 2.4 (taken from paper [20]). The structure of a N-search pattern module with parallelism P = 4. Each of the P copies of the decoder generates the equality signals for C characters, where C is the number of distinct characters that appear in the N search strings. A shared network of SRL16 shift registers provides the results in the desired timing, and P AND gates provide the match signals for each search pattern.

2.5 Pattern partitioning algorithms

To identify which search patterns should be included in a group we have to determine the relative cost of the various different possible groupings. The goal of the partitioning algorithm is (i) to minimize the total number of distinct characters that need to be decoded for each group, and (ii) to maximize the number of characters that appear in the same position in multiple of search patterns of the group (in order to share the shift registers).

The proposed algorithm by Sourdis and Pnevmatikatos implements a simple heuristic and does not guarantee an optimal partitioning of the search patterns. In the next chapter we will discuss this pattern partitioning algorithm in more detail. **The main idea behind this thesis method is that a more "sophisticated" algorithm can be used to achieve a better partitioning of the entire rules set**. A method called **graph partitioning** is proposed, in which a better partition criterion seeks a small cut that partitions the vertices into roughly equal-sized pieces.

Unfortunately, our data sets are not so regular in structure, thereby necessitating more sophisticated partitioning methods. Our problem is to identify which search patterns should be included within a group. Since patterns should be divided evenly across a group set while minimizing the total number of distinct characters that need to be decoded for each group (edges that straddle two subsets), it can be phrased as a graph partitioning problem in which the number of partitions is equal to the number of groups.

2.6 Cost model

We will first present the evaluation on the basic performance and cost of DCAMs. Area cost is calculated in terms of occupied logic cells needed for a design that stores a certain number of matching characters (area cost = logic cells/character).

Figure 2.5 plots the performance both in terms of operating frequency, as well as in processing throughput (Gbps) for the three group sizes (64, 128, 256 rules per group) and for rule sets with sizes between 4,000 and 18,000 total characters. We present these results to show the performance improvement that is achieved by the partitioning technique and not to do a straightforward comparison with the current work. In the next

sections of this thesis an overall comparison between DCAM with the greedy algorithm partitioning and our approach with the graph partitioning technique will be presented.

We can see that all the different designs achieve operating frequencies between 335 and 385MHz. This corresponds to a processing bandwidth between 2.7 and 3.1 Gbps. From these results we can draw two general conclusions for the group size.

- (i) smaller group sizes are more insensitive to the total design size
- (ii) when the group size approaches 256 the performance deteriorates, indicating that optimal group sizes will be in the 64-128 range.



Figure 2.5 (taken from [20]). DCAM Performance of previous work in terms of operating frequency and throughput for the group sizes of 64, 128, and 256 rules and for rule sets with sizes 4.000 and 18.000 characters.

The area cost was also measured and the number of logic cells needed for each search pattern character was plotted in Figure 2.6. As expected, larger group sizes result in smaller area cost due to the smaller replication of comparators in the different groups. Similar to performance, the area cost sensitivity to total rule set size increases with group size. In all, the area cost for the entire Snort ruleset is about 1.28, 1.1 and 0.97 logic cells per search pattern character for group sizes of 64, 128 and 256 rules respectively. While smaller group sizes offer the best performance, it appeared that if the area cost is also taken into account, the medium group size (128) can be considered optimal.



Figure 2.6. (taken from [20])DCAM Area cost of previous work, in terms of operating frequency and throughput for the group sizes of 64, 128, and 256 rules.

Chapter 3

3.1 Pattern Partitioning Algorithms

To identify which search patterns should be included in a group we have to determine the relative cost of the various different possible groupings. A simple algorithm was proposed by Sourdis and Pnevmatikatos that implements a simple heuristic and does not guarantee an optimal partitioning of the search patterns. The cost was computed by finding the set difference between the set of characters used already by the group and the set of characters in the search pattern under consideration. They iterated among all groups and all search patterns until all the patterns have been assigned to a group. This algorithm is described in three steps :

1. First an array is created with one entry for each search pattern. Each array entry contains the set of distinct characters in the search string.

2. Starting with a number of empty partitions or groups, a step of initial assignment of search patterns is performed, to obtain a "seed" pattern in each group of the different groups.

3. Then an iterative method is used: for each group they select an unassigned search pattern so that the cost of adding it to the group is the least among the unassigned patterns. The cost is computed by finding the set difference between the set of characters used already by the group and the set of characters in the search pattern under consideration. Iteration is made among all groups and all search patterns until all the patterns have been assigned to a group.

The algorithm was also compared with a straightforward approach of just sorting the search patterns, and it was observed that using the group identified by their algorithm the area cost was about 5% smaller and 5% faster than the one using partitioning based on sorted search patterns. This algorithm was more efficient in minimizing the number of shift registers requiring 9% fewer shift registers than the sorting the search patterns. For the entire SNORT rule set and using 24 groups, the algorithm produced groups that contain an average of 54 distinct search characters each. Therefore each of the decoders is significantly smaller that a full 8-to-256 decoder.

This work proposes a partitioning alternative. The partitioning algorithm of Sourdis & Pnevmatikatos is greedy, and hence may leave room for further improvements. A more sophisticated algorithm could take into account the exact location of the similarities between search patterns (in order to increase the degree of shift register sharing), and would use a global instead of local approach to cost minimization.

3.2 Graph Partitioning Algorithms

3.2.1 Problem Formulation

A complete NIDS based on Snort's rules requires a system optimized for hundreds of rules, many of which require string matching against the entire data segment of a packet. Snort, the open-source IDS has thousands of content-based rules. Each of these rules require that a packet be searched in its entirety for the occurrence of some "fingerprint" string. Using naive methods, this is unworkable. Using more sophisticated algorithms or higher levels of parallelism it becomes tenable. Most research in this area has achieved to develope hardware architectures that can handle more than a few hundred rules at reasonable speeds.

The goal is to partition the entire set of search patterns in smaller groups. Some data are easy to distribute. For example, dense array-based problems typically have a high degree of regularity, allowing the array elements to be distributed using straightforward blocked, cyclic, or block-cyclic schemes. These distributions are advantageous due to their simplicity and their ability to take advantage of an array's regular structure. **Unfortunately, our data sets are not so regular in structure, thereby necessitating more sophisticated partitioning methods. Our problem is to identify which search patterns should be included within a group.**

Since patterns should be divided evenly across a group set while minimizing the total number of distinct characters that need to be decoded for each group (edges that

straddle two subsets), it can be phrased as a graph partitioning problem in which the number of partitions is equal to the number of groups.

3.3 Previous work on Graph Partitioning

Graph Partitioning is a very common problem and has a large number of applications such as circuit layout, compiler design, and load balancing.

A graph partitioning problem in its most general form, requires dividing the set of nodes of a weighted graph into k disjoint subsets or partitions such that the sum of weights of nodes in each subset is nearly the same (within a user-supplied tolerance) and the total weight of all of the edges connecting nodes in different partitions is minimized.



Figure 3.1: Graph partitioning example

Several different flavours of graph partitioning arise depending on the desired objective function:

- **Minimum cut** The smallest set of edges to cut that will disconnect a graph can be efficiently found using network flow methods.
- **Graph partitioning** A better partition criterion seeks a small cut that partitions the vertices into roughly equal-sized pieces. Unfortunately, this problem is NP-complete. Fortunately, heuristics discussed below work well in practice. These heuristics often can handle rather large graphs with more than a million vertices and deliver good solutions. In contrast to the development of heuristics only a little expense has been done in the development of exact algorithms. From the NP-hardness fact it is clear that generally only relatively small graphs can be solved exactly. Nevertheless, exact solutions are of interest for applications and for the validation of heuristics.

Maximum cut - Given an electronic circuit specified by a graph, the maximum cut defines the largest amount of data communication that can simultaneously take place in the circuit. The highest-speed communications channel should thus span the vertex partition defined by the maximum edge cut. Finding the maximum cut in a graph is NP-complete, despite the existence of algorithms for minimum cut. However, heuristics similar to those of graph partitioning can work well.

Unfortunately, graph partitioning is a NP-hard problem, and therefore all known algorithms for generating partitions merely return approximations to the optimal solution. In spite of this theoretical limitation, numerous algorithms for graph partitioning have been developed that generate high-quality partitions in very little time.

Spectral partitioning methods (Figure 3.2) are known to produce good partitions for a wide class of problems, and they are used quite extensively [12]. However, these methods are very expensive since they require the computation of the eigenvector corresponding to the second smallest eigenvalue (Fiedler vector). Another class of graph partitioning techniques uses the geometric information of the graph to find a good partition.



Figure 3.2: Spectral Methods. (1) The Laplacian matrix LG of the graph is = A -D (the adjacency matrix – the degree matrix) (2) Compute the second eigenvector (Fiedler vector) of LG. (3) The Fiedler vector associates a value with each vertex, this value is used to order the vertices and the list is split in half.

Geometric partitioning algorithms [13,14,15] tend to be fast but often yield partitions that are worse than those obtained by spectral methods. Among the most prominent of these schemes is the algorithm described in [13,14]. This algorithm produces partitions that are provably within the bounds that exist for some special classes of graphs (that includes graphs arising in finite element applications). However, due to the randomized nature of these algorithms, multiple trials are often required (5 to 50) to obtain solutions that are comparable in quality to spectral methods. Geometric graph partitioning algorithms are applicable only if coordinates are available for the vertices of the graph. In many problem areas (e.g. linear programming, VLSI), there is no geometry associated with the graph.

Another class of graph partitioning algorithms reduces the size of the graph (i.e. coarsen the graph) by collapsing vertices and edges, partition the smaller graph, and then uncoarsen it to construct a partition for the original graph. These are called **multilevel graph partitioning schemes**. Some researchers investigated multilevel schemes primarily to decrease the partitioning time at the cost of somewhat worse partition quality [43]. Recently, a number of multilevel algorithms have been proposed [16,17,18,] that further refine the partition during the uncoarsening phase. These schemes tend to give good partitions at a reasonable cost. Bui and Jones [16] use random maximal matching to successively coarsen the graph down to a few hundred vertices, they
partition the smallest graph and then uncoarsen the graph level by level, applying Kernighan-Lin to refine the partition (see Figure 3.3).



Figure 3.3: Given a graph that has been partitioned (sub-optimally), improve the partition maintaining load balance, repeatedly find a pair of vertices, one from each subdomain and swap their subdomains. At each iteration the algorithm swaps subsets consisting of equal number of vertices between the two sets to reduce the number of edges joining the two sets. The algorithm terminates when it is no longer possible to reduce the number of edges by swapping subsets, or when a specified number of swaps have been made.

Hendrickson and Leland [5] enhance this approach by using edge and vertex weights to capture the collapsing of the vertex and edges. In particular, this latter work showed that multilevel schemes can provide better partitions than spectral methods at lower cost for a variety of finite element problems. **In this thesis the tool used for the implementation is based on the work of Hendrickson and Leland**. The algorithm studied here is called multilevel recursive bisection [8], which we will describe latter in this chapter.

3.3.1 P-way Partition

In a graph-based form, each node represents a search pattern while each edge represents a data dependence between two vertices. In this thesis, a graph G=(V,E) is

defined in terms of a set of vertices V , and a set of edges E. Edges connect vertices from V pair-wise and are undirected. Self-loops are not permitted.

A *p*-way partition of a graph is a mapping $\mathcal{P}: \mathcal{V} \rightarrow [1...p]$ of its vertices into *p* subsets $S_1, S_2, ..., S_p$. Every partition generates a set of **cut edges** E_c defined as the subset of \mathcal{E} whose endpoints lie in distinct partitions. The **weight** of each subset, $|S_i|$ is defined to be the number of vertices mapped to that subset by \mathcal{P} .

The efficient implementation of many parallel algorithms usually requires the solution to a graph partitioning problem, where vertices represent computational tasks and edges represent data exchanges. Depending on the amount of the computation performed by each task, the vertices are assigned a proportional weight. Similarly, the edges are assigned weights that reflect the amount of data that needs to be exchanged. A k-way partitioning of this computation graph can be used to assign patterns to k groups. Since the partitioning assigns to each patterns tasks whose total weight is the same, the work is balanced among k groups. Furthermore, since the algorithm minimizes the edge-cut (subject to the balanced load requirements), the communication overhead is also minimized.

3.3.2 Recursive Bisection

An instance of graph partitioning that deserves special attention is the **graph bisection problem**. This is simply a variation on graph partitioning in which the graph G must be divided into two subsets. Although bisection seems considerably easier than general p-way partitioning, it is still NP-hard.

Most p-way partitioning algorithms utilize a divide-and-conquer approach known as **recursive bisection.** This technique generates a p-way partition by performing a bisection on the original graph and then recursively considering the resulting subgraphs (**Figure 3.5**). It has been shown that even if recursive bisection is performed using an optimal bisection algorithm, it can still result in a suboptimal p-way partition [11]. In spite of this theoretical limitation, recursive bisection remains the primary graph partitioning strategy due to its simplicity compared to computing p-way partitions directly.



Figure 3.4: An example demonstrating the use of recursive bisection to compute an eight-way partition for an abstract graph.

3.3.3 Multilevel Techniques

One recent approach that has greatly accelerated the partitioning of graphs is the use of **multilevel techniques**. These techniques are analogous to multigrid methods for solving numerical problems. Both approaches construct a hierarchy of approximations to the original problem so that a coarse solution can quickly be generated. This solution is then progressively refined at the more detailed levels of the hierarchy until a solution for the original problem is reached. In the context of graph partitioning, this translates into creating a simplified graph that approximates the input graph, finding a partition for it, and then refining that partition to create a partition for the original graph.



Figure 3.5: A schematic of the multilevel technique. The original graph (bottom left) undergoes a series of coarsening steps that reduce it to a smaller graph. This coarsest graph is partitioned using a standard algorithm. The partition is then propagated down to the finer graphs, potentially refining it at each level to account for the additional degrees of freedom. The result is a partition for the original graph.

All multilevel techniques for graph partitioning share the same general computational structure, though the details may vary:

Coarsen: Given the input graph G₀ = (V₀, E₀) construct a series of increasingly smaller graphs G_i = (V_i, E_i) consisting of G₁, G₂,...,G_m graphs, such that |V₀| > |V₁| >...> |V_m|. During the coarsening phase, a sequence of smaller graphs, each with fewer vertices, is constructed. Graph coarsening can be achieved in various ways. Some possibilities are shown in Figure 3.6.



Figure 3.6: Different ways to coarsen a graph

- Partition: A two-way partition of the graph G_m is computed, that partitions V_m into two parts, each containing half the vertices of G_0 , using a standard algorithm.
- Uncoarsen: Propagate the solution for G_m down to the finer graphs, potentially refining it at each level. In other words, the partition P_m of G_m is is projected back to G_0 by going through intermediate partitions $P_{m-1}, P_{m-2}, ..., P_1, P_0$.

This process results in a partition for the original graph (Figure 3.5). The hope is that multilevel techniques will reduce the time required to compute partitions without sacrificing quality. In practice, the use of multilevel techniques has proven not only to accelerate partition generation, but also to produce better partitions than traditional single level techniques [5].

3.4 Our Approach

Our problem is to identify which search patterns should be included in a group. In our approach a node represents a string. Our strings are a sequence of hexadecimal characters like the ones presented below:

00010003000100 00012f 000143 000186A0 202F2525 202f485454502f312e 2041555448454e544943415445207b

These strings derived from a typical Snort's ruleset after converting the text characters to their ASCII code. All characters are in a single text file, one string per line. We construct a graph that each node represents a single string. So if a file consists of 100 lines, our graph will have 100 nodes.

An edge (weighted) exists between two nodes in this graph, whenever the following rule applies : the first and the second digits of two strings are the same. This is because the characters in the rule set are given in hexadecimal notation, where a character is represented by two digits.

So in the next example in **Figure 3.7** we see a match in the first and the second string and this is the "00" digits, also the next characters form a match, the "01" digits.



Figure 3.7 : Two matching rules

There is no other match in this example. So the two strings were "matched" two times, so 2 will be the **weight** in our graph construction algorithm.

Our simple graph construction algorithm, builds the graph according to this "matching" rule. It takes the first line of the rule's file and compares it to all the other lines. Whenever a rule-match between two strings (nodes) is found, an edge is added to the graph between the two matching nodes. The weight of each edge is also included. Then the next line is compared with all the other lines, and so on until the entire file is processed. The graph is written in file in the format that is required by the software partitioning software (METIS), which will be described in more detail in the next chapter.

We used *multilevel recursive bisection,* an algorithm implemented in the METIS software package for partitioning our graph. The result we get after we apply the partitioning algorithm is a simple file with :

- number of lines=number of nodes (=number of strings of the input file)
- each lines has a single number in the range 0-5 (for a 6-way partition)
- each line in the resulted file indicates in which partition each node (string) belongs, like the example we show in the next table. The first column is the result of METIS while the second column shows the associated string in the partition number if we check the original rule's file.

Partition number that rule belongs to	String (rule) assigned
(METIS file)	
3	0A0000018504000080726F6F7400
5	0A2020202020
4	0a433a6461656d6f6e0a52
5	0a43726f6f740a4d70726f67
5	0a43726f6f740d0a4d70726f67
5	0a442f
0	0A68656c700A71756974650A

Table 3.1 : This is a simple METIS output file example

3.5 Example: How to coarsen a Graph

In most coarsening schemes, a set of vertices of G_i is combined to form a single vertex of the next level coarser graph G_{i+1} . This edge collapsing idea can be formally defined in terms of matchings. A matching of a graph is a set of edges, no two of which are incident on the same vertex. Thus, the next level coarser graph G_{i+1} is constructed from G_i by finding a matching of G_i and collapsing the vertices being matched into multinodes. The edges in this set (matching set) are removed, and the two nodes connected by an edge in the matching are collapsed into a single node whose weight is the sum of the weights of the component nodes. The unmatched vertices are simply copied over to G_{i+1} . Since the goal of collapsing vertices using matchings is to decrease the size of the Graph G_i the matching should be maximal. A matching is called maximal matching if it is not possible to add any other edge to it without making two edges become incident on the same vertex. Note that depending on how matchings are computed, the size of the maximal matching may be different. The coarsening phase ends when the coarsest graph G_m has a small number of vertices or if the reduction in the size of successively coarser graphs becomes too small. An example of the coarsening phase in detail is given in Figure 3.8 (a,b,c,d).

Given a weighted graph after any stage of coarsening, there are several choices of matchings for the next coarsening step. A simple matching scheme [5] known as **random matching (RM)** randomly chooses pairs of connected unmatched nodes to include in the matching. In [7], Karypis and Kumar describe a heuristic known as **heavy-edge matching (HEM)** to aid in the selection of a matching that not only reduces the run time of the refinement component of graph partitioning, but also tends to generate partitions with small separators. The strategy is to randomly pick an unmatched node, select the edge with the highest weight among the edges incident on this vertex that connect it to other unmatched vertices, and mark both vertices connected by this edge as matched. Note that the weight of an edge connecting two nodes in a coarsened version of the graph is the number of edges in the original graph that connect the two sets of original nodes collapsed into the two coarse nodes. HEM, by absorbing the heavier edges, generates coarse graphs whose nodes are loosely connected (by the lighter

remaining edges), thus ensuring that a partition of the coarse graph corresponds to a good partition of the original graph.



Figure 3.8.a : Original Graph and a matching (a set of edges, no two of which are incident on the same vertex). We assume node-edges weights equal to 1 at this example for simplicity.



Figure 3.8.b: Graph after one step of coarsening. The edges in the matching set are removed, and the two nodes connected by an edge in the matching are collapsed into a single node whose weight is the sum of the weights of the component nodes. (Big nodes represent the connected nodes from the previous coarsening phase). Taking into account edge weights also, the weight of the new edge is the sum of the weights of the edges "collapsed" into it.



Figure 3.8.c: Next step is the matching in the coarse graph



Figure 3.8.d: Graph after the two steps of coarsening.

3.6 Example: How to Partition a Graph

One way to get the initial k-way partitioninig is to keep coarsening the graph until it contains exactly k nodes. This coarse k-node graph serves as a good initial partitioning. There are two problems with this approach. First, for many graphs, the reduction in the size of the graph in each coarsening step becomes very small after some coarsening steps, making it very expensive to continue with the coarsening process. Second, even if we are able to coarsen the graph down to only k nodes, the weights of these nodes are likely to be quite different, making the initial partitioning highly unbalanced. **In METIS the partitioning phase is done using the multilevel bisection algorithm.**

In this second phase of the multilevel partitioning algorithm computes a high quality bisection P_m of the coarse graph G_m , that partitions V_m (vertices) into two parts,

such that each part contains roughly half of the vertex weight of the original graph G_0 , using a standard algorithm (like spectral partitioning methods, geometric methods, multilevel bisection, p-way partition, already discussed before in this chapter). Since during coarsening the weights of the vertices and edges of the coarser graph were set to reflect the weights of the vertices and edges of the finer graph, G_m contains sufficient information to intelligently enforce the balanced partition and the small edge-cut requirements.

We will not get into details on this phase since there are many algorithms that compute the multilevel bisection, most of them are combinations of other algorithms and are not useful in this thesis. Here we will take the simplest rule to find an initial partition and this is the **small edge-cut**, that is to minimize the number of edges crossing the cut.

3.7 Example: How to Uncoarsen a Graph

In this third phase of the multilevel partitioning algorithm the goal is to "redraw" the original graph. In other words the partition P_m of the coarser graph G_m is projected back to G_0 by going through the graphs $G_{m-1}, G_{m-2}, ..., G_1$.

After projecting a partition, a refinement algorithm is used to select two subsets of vertices, one from each part such that when swapped the resulting partition has a smaller edge-cut. Here we will continue our example with no refinement procedure. In **Figure 3.9** we give an example of the uncoarsening technique.



Figure 3.9: The uncoarsening phase of the graph. Note that is just a simple example, no heuristics were taken into the initial graph partition and no refinement procedure is made at this point.

Chapter 4

METIS: A Software Package for Partitioning Unstructured Graphs

4.1 Introduction

Algorithms that find a good partitioning of highly unstructured graphs are critical for developing efficient solutions for a wide range of problems in many application areas on both serial and parallel computers. For example, large-scale numerical simulations on parallel computers, such as those based on finite element methods, require the distribution of the finite element mesh to the processors. This distribution must be done (i) so that the number of elements assigned to each processor is the same, and (ii) the number of adjacent elements assigned to different processors is minimized.

The goal of the first condition is to balance the computations among the processors. The goal of the second condition is to minimize the communication resulting from the placement of adjacent elements to different processors. Graph partitioning can be used to successfully satisfy these conditions by first modeling the finite element mesh by a graph, and then partitioning it into equal parts.

4.2 What is METIS

METIS [30] is a software package for partitioning large irregular graphs, partitioning large meshes, and computing fill reducing orderings of sparse matrices. The algorithms in METIS are based on multilevel graph partitioning described in [6], [7], [8]. The advantages of METIS is that it's very fast and it provides good quality partitions compared to other similar software packages.

4.3 METIS's Stand-Alone Programs

METIS provides a variety of programs that can be used to partition graphs, partition meshes, compute fill-reducing orderings of sparse matrices, as well as programs to convert meshes into graphs appropriate for METIS's graph partitioning programs.

The rest of this section provides detailed descriptions about the functionality of these programs, how to use them, the format of the input files required by them, and the format of the produced output files.

4.4 Graph Partitioning Programs provided by METIS

METIS provides two programs pmetis and kmetis for partitioning an unstructured graph into *k* equal size parts. The partitioning algorithm used by **pmetis** is based on *multilevel recursive bisection*, whereas the partitioning algorithm used by **kmetis** is based on *multilevel k-way partitioning*. Both of these programs are able to produce high quality partitions. However, depending on the application, one program may be preferable than the other. In general as concluded from METIS documentation, kmetis is preferred when it is necessary to partition graphs into more than eight partitions. For such cases, kmetis is considerably faster than pmetis.

On the other hand, pmetis is preferable for partitioning a graph into a small number of partitions. Both pmetis and kmetis are invoked by providing two arguments at the command line as follows:

pmetis GraphFile Nparts *kmetis* GraphFile Nparts

The first argument, *GraphFile*, is the name of the file that stores the graph, while the second argument, *Nparts*, is the number of partitions that is desired. Both pmetis and kmetis can partition a graph into an arbitrary number of partitions. Upon successful execution, both programs display statistics regarding the quality of the computed partitioning and the amount of time taken to perform the partitioning. The actual partitioning is stored in a file named GraphFile.part.Nparts, whose format is described later in this section.

Figure 4.1 taken from METIS documentation, shows the output of pmetis and kmetis for partitioning a graph into 100 parts. From this figure we see that both programs initially print information about the graph, such as its name, the number of vertices (#Vertices), the number of edges (#Edges), and also the number of desired partitions (#Parts). Next, they print some information regarding the quality of the partitioning. Specifically, they report the number of edges being cut (Edge-Cut) by the partitioning, as well as the balance of the partitioning1. Finally, both pmetis and kmetis show the time taken by the various phases of the algorithm. All times are in seconds.

```
prompt% pmetis brack2.graph 100
METIS 4.0 Copyright 1998, Regents of the University of Minnesota
Graph Information -----
 Name: brack2.graph, #Vertices: 62631, #Edges: 366559, #Parts: 100
Recursive Partitioning...
 100-way Edge-Cut: 37494, Balance: 1.00
Timing Information -----
 I/0:
                   0.820
 Partitioning:
                   6.110
                        (PMETIS time)
 Total:
                   6.940
prompt% kmetis brack2.graph 100
                 METIS 4.0 Copyright 1998, Regents of the University of Minnesota
Graph Information -----
 Name: brack2.graph, #Vertices: 62631, #Edges: 366559, #Parts: 100
K-way Partitioning... -----
 100-way Edge-Cut:
           37310, Balance: 1.03
Timing Information -----
 I/0:
                   0.820
 Partitioning:
                   1.750
                        (KMETIS time)
                   2.570
 Total:
```

Figure 4.1: Output of pmetis and kmetis for 100-way partition (taken from METIS documentation [30])

4.5 Graph Checker

METIS provides a program called **graphchk** to check whether or not the format of a graph is appropriate for use with METIS. This program should be used whenever there is any doubt about the format of any graph file. It is invoked by providing one argument at the command line as follows:

graphchk GraphFile

where *GraphFile* is the name of the file that stores the graph. After creating the graph file in the format required by METIS, we used graphchk command to verify that the graph is consistent with the required format.

4.6 Input File Formats

The various programs in METIS require as input a file storing a graph. The format of such a file is described in the following sections.

4.6.1 Graph File

The primary input of the partitioning programs in METIS is the graph to be partitioned or ordered. This graph is stored in a file and is supplied to the various programs as one of the command line parameters. A graph G = (V, E) with n vertices and m edges is stored in a plain text file that contains n + 1 lines (excluding comment lines). The first line contains information about the size and the type of the graph, while the remaining *n* lines contain information for each vertex of G. Any line that starts with "%" is a comment line and is skipped.

The first line contains either two (n, m), three (n, m, fmt), or four (n, m, fmt, ncon) integers. The first two integers (n, m) are the number of vertices and the number of edges, respectively. Note that in determining the number of edges m, an edge between any pair of vertices v and u is counted **only once** and not twice (*i.e.*, we do not count the

edge (v, u) separately from (u, v). For example, the graph in **Figure 4.4** contains 11 vertices. The third integer (fmt) is used to specify whether or not the graph has weights associated with its vertices, its edges, or both.

Table 4.1 describes the possible values of *fmt* and their meaning. Note that if the graph is unweighted (*i.e.*, all vertices and edges have the same weight), then the *fmt* parameter can be omitted. Finally, the fourth integer (*ncon*) is used to specify the number of weights associated with each vertex of the graph but this is not used in our work.

fmt	Meaning
0	The graph has no weights associated with either the edges or the vertices
1	The graph has weights associated with the edges
10	The graph has weights associated with the vertices
11	The graph has weights associated with both the edges & vertices

Table 4.1: The various possible values for the fmt parameter and their meanings

The remaining n lines in the file store information about the actual structure of the graph. In particular, the i th line (excluding comment lines) contains information that is relevant to the i th vertex. In the remaining of this section we illustrate this format by a sequence of examples. Note that the vertices are numbered starting from 1. Furthermore, the vertex-weights must be integers greater or equal to 0, whereas the edge-weights must be strictly greater than 0. The simplest format for a graph G is when the weight of all vertices and the weight of all the edges is the same.

This format is illustrated in **Figure 4.4(a)**. Note, the optional *fmt* parameter is skipped in this case. However, there are cases in which the edges in G have different weights. This is accommodated as shown in **Figure 4.4(b)**. Now, the adjacency list of each vertex contains the weight of the edges in addition to the vertices that is connected with. If v has k vertices adjacent to it, then the line for v in the graph file contains 2 * k numbers, each pair of numbers stores the vertex that v is connected to, and the weight of

the edge. Note that the *fmt* parameter is equal to 1, indicating the fact that G has weights on the edges.



Figure 4.4: Storage format for various type of graphs (figure taken from METIS documentation [30])

Let's explain these file formats shown in the above figure in more detail. A graph must be stored in an adjacency matrix representation as stated above. So if we look at the graph file of graph (a) in **Figure 4.4**, we see that the first line contains the number of nodes (7) and the number of edges (11, and then the file goes like this :

- node number 1 is connected with nodes 5 3 2
- node number 2 is connected with nodes 1 3 4

```
and so on .
```

If the graph has weights associated with edges, then the file should be interpreted like this:

node 1 is connected

• with node 5 with weight 1

- with node 3 with weight 2
- with node 2 with weight 1

so if we put all this in one line we get the line of the graph file for node 1 which is 5 1 3 2 2 1.

Our simple graph composition algorithm first counts the number of the nodes and also initializes a variable to count the total number of the edges in the graph. The algorithm builds the graph according to the "matching" rule. A match, as we have already mentioned, is encountered whenever the first and the next character of the two strings are the same. So this simple algorithm takes the first line of the rule's file and compares it to all the other lines.

Whenever a rule-match between two strings (nodes) is found, the line number of the matching node is added to the line of the node examined, a number indicating the weight of that match and also the number that holds the edges totality is updated. Then the next line is compared with all the other lines, and so on until the entire rule's file is progressed.

4.6.2 Output File Formats

The output of METIS is either a partition or an ordering file, depending on whether METIS is used for graph/mesh partitioning or for sparse matrix ordering. For our graph partitioning the output is a partition file.

The partition file of a graph with n vertices in METIS, consists of n lines with a single number per line. The i th line of the file contains the partition number that the i th vertex belongs to. Partition numbers start from 0 up to the number of partitions minus one.

After constructing the file that represents our graph in the adjacency list representation required by METIS, we run the *graphchk* utility to verify that our graph is in the correct format. This is a simple output we got:

Program graphchk

Graph Information -----

Name: graph_all, #Vertices: 1474, #Edges: 297009

Checking Graph... -----

The format of the graph is correct!

This utility notifies you if the format of the graph is not correct, especially if the number of edges specified is not correct. The program tells you the number of edges it encountered so that you can correct your graph file.

Then we used *pmetis* program to partition our graph into 6 groups . This is the output we got by this procedure.

METIS 4.0.1 Copyright 1998, Regents of the University of Minnesota

Graph Information ------Name: graph_all, #Vertices: 1474, #Edges: 297009, #Parts: 6

Recursive Partitioning... -----

6-way Edge-Cut: 258510, Balance: 1.01

Timing Information	1
I/O:	0.100
Partitioning:	0.110 (PMETIS time)
Total:	0.210

The next step was to partition the graph into 12 groups.

As we can see from the above print-outs of the program, the main advantage of using METIS software package is that is it very quick and straightforward to obtain the required partitions.

4.7 Graph visualization

We tried several tools in order to visualize our initial graph (as explained in section 3.4). These were Dotty, Pajek and Tulip. Unfortunately, due to the fact that **our graph contains 1475 nodes and 297024 edges**, it is very difficult to see how the graph actually is represented. Tulip was the only program that could handle such a graph since both Dotty and Pajek resulted in errors. Tulip input file format is a little bit complex, but the main advantage of this tool is that it can accept other input file formats too. The most simple file format acceptable by Tulip is the "dot" format required by the Dotty

software tool, which is presented in a simple example below. So we created a file that described our graph in the dot format and then gave this file as import file in Tulip.

```
digraph G {
edge [arrowhead="none", arrowtail="none"]
a -> b [label="5", style=dashed];
a -> c [label="7",];
c -> d; b -> c; d -> a;
}
```

Figure 4.5 A simple dot file



Figure 4.6: The corresponding graph from the above dot file

In the next figures we can see how the graph actually looks like. As we have already said, our graph has too many edges and nodes, so we cannot actually distinguish in **Figure 4.7** the nodes from the edges. We zoom with the tool in various areas of the graph to take a better look **(Figures 4.8, 4.9, 4.10)**.

7413 1094 17 1000 2 78 A
<u>ັ້ງປຽງ</u> ງ 100 84 2 284 ັ
^{ିମ} _{ମସମ} ୨୦୦ ^{ଅସ}
8 1858 461 50 4 1402 52
9077 983 2 1 2 830 18 935
⁰⁰ 82 0 1829 1 27 203 1
1000 1000
<u>3000</u> <u>19</u> 10 29 <u>390</u>
778 8 1 8 7 60 55 1 508 0 1003
ີ <u>36 08 ງ</u> ງງ ງ088 ດອງ ວັງ
488 28 1181
10 1008 1 ⁹⁷⁰ 1 2 989 702
496 _ 1 78 ²¹ 70 1146 ⁸⁴
80 1 22 ° 80 0 220 97
5 210 239 1 37

Figure 4.7: Graph from Tulip



Figure 4.8 : A zoom on an area of the graph



Figure 4.9: Another zoom



Figure 4.10: Nodes have too many edges

Chapter 5

5.1 Results and Evaluation

5.1.1 Introduction

This thesis explores **partitioning algorithms** to increase the character's sharing in a DCAM architecture in order to reduce the area cost of character matching.

To identify which search patterns should be included in a group we have to determine the relative cost of the various different possible groupings. The goal of the partitioning algorithm is

- to minimize the total number of distinct characters that need to be decoded for each group (edges that straddle two subsets)
- to maximize the number of characters that appear in the same position in multiple copies of search patterns of the group (in order to share the shift registers)

We used the official SNORT rule set [38] which consists of a total of

about 1,500 rules and a corresponding 18,000 characters. The proposed partitioning algorithms by Sourdis & Pnevmatikatos, take into account the exact location of the similarities between search patterns (in order to increase the degree of shift register sharing) and uses a global instead of local approach to cost minimization.

In the previous work of Sourdis, for the evaluation of the impact of partitioning on his proposed architecture, he considered three different group sizes: 64,128 and 256 rules per group. Experimentally he observed that groups smaller than 64 or larger than 256 rules are inefficient and that the range 64-256 is sufficient to explore grouping efficiency.

5.2 Results

Our partitioning scheme, which is based in graph partitioning techniques described in chapter 3, partitions the rule set into 6 and 12 groups. The number of rules in each group is shown in the following tables.

Part.	Distinct	Total	Size of	
number	Chars	number	max	
	/per	of	string	
	partition	strings	(chars)	
1	86	244	39	
2	67	244	37	
3	68	244	39	
4	202	243	34	
5	92	245	32	
6	152	243	38	

Table 5.1: Number of distinct characters per partition, if we partition into 6 groups

Part.	Distinct	Total	Size of	
number	Chars	number	max	
	/per	of	string	
	partition	strings	(chars)	
1	59	122	34	
2	61	122	31	
3	50	123	37	
4	77	121	39	
5	65	122	39	
6	64	122	37	
7	143	121	34	
8	157	122	26	
9	122	122	38	
10	103	122	25	
11	81	123	31	
12	77	121	32	

 Table 5.2 Number of distinct characters per partition, if we partition into 12 groups

We measured performance in terms of operating frequency, and throughput (*throughput = frequency x input bits*) and compared our results with the ones that were generinted the former work of Sourdis. In our case the input bit per cycle is 8. We also

measured area cost and calculated the number of logic cells needed for each search pattern character and we present the total number of Flip-Flops, the number of LUTs and the number of SRL16s in Tables 5.3 and 5.4. Logic cells are the fundamental element of both Altera's and Xilinx's devices (A logic cell consists of a 4-input LUT and a flip-flop (plus carry chain logic etc.) Two *Logic Cells* form one *Slice*, and therefore, it is the most proper measure for evaluating the area cost of a design.

For the 6-group partition (average number of rules in each group is 244) in Xilinx Virtex2 we achieved operating frequency of 366MHz. Sourdis has reported an operating frequency of 331MHz for his 6 partition. For the 12-group partition (average number of rules in each group is 122) in the Virtex2 device, we achieved operating frequency of 310MHZ, while Sourdis has reported 317MHz frequency for his 12 partition. These results are better seen in the following tables

6-way partition	Our Work	Sourdis Work
SLOW Clock	2.767ns =361Mhz	2.812ns=355MHz
FAST Clock	2.732ns= 366MHz	3.014ns=331Mhz

12-way partition	Our Work	Sourdis Work
SLOW Clock	2.205ns =453Mhz	2.451ns=407MHz
FAST Clock	3.219ns= 310MHz	3.149ns=317Mhz

In terms of performance, after giving our partitions for simulation in Virtex2 machine (Xilinx's device), we can see that all the different partitions achieve operating frequencies between 310 and 453MHz for Virtex2. The results show that while for the smallest rule set (12 partitions) both implementations operate at similar frequencies with our method giving 2480 Gbps and Sourdis method with 2536 Gbps, when the rule set size increases (6 partitions) , the scalability of the our approach is better and achieves about 10% better frequency.

Unlike performance, the effect of group size on the area cost is more pronounced. As expected, larger group sizes result in smaller area cost due to the smaller replication of comparators in the different groups. Therefore, designs that are partitioned in many groups have higher area cost as compared to with fewer partitions Similar to performance, the area cost sensitivity to total rule set size increases with group size.

The results present next are all calculated into the Virtex2 device. The number of logic cells is calculated form the number of slices. Each slice is 2 logic cells and this is what we call "Number of occupied Slices". The number of LUTs is the "Total Number 4 input LUTs". The Flip-Flops are "Number of Slice Flip Flops". The SRL16 are the shift registers that delay the decoded data. This is basically what states the number of unique characters per position, and we call this "Number used as Shift registers".

By comparing our results with the ones generated by Sourdis, we can see that there are slight differences, especially in the number of shift registers. Our partitioning did a better work in terms of grouping (number of different characters per position). As for the other differences encountered, these are more or less around 1%. Our partitioning into 6 groups (average number of rules 245) demands more logic cells than the one of Sourdis. On the contrary when we partition with an average number of rules of 122-124 we get a smaller number of logic cells.

6-group partition	slices	Logic cells (slices*2)	LUTs	Flip-Flops	SRL16s
Our partition	8616	17232	13816	15552	4218
Sourdis partition	8567	17134	13946	15677	4309

 Table 5.3 : Comparison in the 6 partition groups

12-group partition	slices	Logic cells (slices*2)	LUTs	Flip-Flops	SRL16s
Our partition	9458	18916	15828	17808	5301
Sourdis partition	9577	19154	15735	17866	5391

Table 5.4: Comparison in the 12 partition groups

From the above Tables (5.3 and 5.4) we see some peculiar results. In the 6 partitioning case Sourdis has reported less Slices while the number of Flip-Flops and LUTs are bigger. This is a weird situation, probably has to do with the tool used to get these results.

Finally we collect the results for area cost and put them all in a comparison chart shown in the next figure.



Figure 5.1: A comparison in the area cost of both implementations (1) From both graphs and both implementation we can see that partitioning designs in smaller groups increases performance, but also increases area cost. (2) Comparing the two approaches, we managed to lower the area cost in smaller groups(that is in 12 partitions), but in the increased group size (6 partitions) our area cost is increased

5.3 Conclusions

Throughout this work we discussed string matching as the major performance bottleneck in intrusion detection systems. We investigated the background work in new string matching micro-architectures and investigated the efficiency of FPGA-based solutions. We first accented the role of string matching in intrusion detection systems. String matching is the most computational intensive part of such systems and limits their performance. Further, we analyzed the set of NIDS patterns, and grouped them with a software-based solution in order to improve the FPGA's performance. Intrusion detection systems running on general purpose processors have limited performance, while on the contrary, ASIC and FPGA-based systems can achieve better performance. In particular, FPGAs offer the flexibility needed in such systems for fast ruleset update.

The preceding work in which this thesis was build on, has shows that FPGAs are well suited for implementing intrusion detection systems, achieving high speed processing in reasonable cost. Our results offer a distinct step forward in the preprocessing needed in the prior work of the DCAM architecture as we implemented the partitioning step (a method to improve the performance of this architecture) with a crisp and complete algorithm-based solution. Finally, our implementation offers simplicity and regularity, and hence it is straightforward to test it with more rulesets and then experiment with the results on the DCAM module in order to design a complete and sophisticated intrusion detection system.

Our partitioning did a better work in terms of grouping (number of different characters per position). As for the other differences encountered, these are more or less around 1%. Our partitioning into 6 groups (average number of rules 245) demands more logic cells than the one of Sourdis while it achieves about 10% better frequency. On the contrary when we partition into 12 groups (smaller rule set) we have a small drop in frequency but the number of logic cells is smaller compared to the ones of Sourdis implementation. This is what we expected since in general we have stated that systems with high performance are costly and systems that have lower cost are at the expense of low performance.

5.4 Discussion and future work

New partition algorithms that rely on combinational, algebraic and geometric ideas have been created and carefully implemented the last years. These contributions make it possible to compute high quality partitions of large graphs in a few minutes on a workstation. Much progress has been made to date and continues to be made at this time in the designing of new algorithms and understanding their behaviour. Graph partitioning has an important role to play in the design of many serial algorithms by means of the divide and conquer paradigm. Some applications include circuit partitioning and layout, VLSI design and Computer-aided design.

Karypis and Kumar, in METIS, build on the multilevel approach with the goal of finding a "best" algorithmic choice for each of the coarsening, partitioning, and refinement stages [6]. Although an exhaustive cross-product of all techniques would be infeasible, their study does an excellent job of examining each stage independently and measuring its impact on the overall partitioning time. Their goal is to find algorithms that represent a good tradeoff between running time and quality. For instance, rather than using a randomized technique for generating maximal matchings, they suggest a strategy called *heavy edge matching* in which edges with higher weight are given priority for inclusion in the matching. The intuition behind this heuristic is that heavier edges will typically be disadvantageous to cut, and therefore collapsing them into a multi-node will remove them from consideration in coarser graphs.

Another interesting result of the Karypis-Kumar study is that using spectral partitioning on the coarsest graph proves not only to be slower than greedier algorithms (as expected), but also results in partitions of significantly worse quality. This indicates that while spectral methods work well on large graphs whose complexity may foil greedier algorithms, their use on smaller graphs may be overkill.

The multilevel suite of choice according to Karypis and Kumar consists of: heavy edge matching for the contraction phase; the greedy graph growing algorithm for the partitioning phase; and a variation of Kernighan-Lin called BKL(*,1) for the refinement phase. BKL(*,1) only considers moving the boundary vertices and uses just a single iteration when the graph becomes sufficiently large (contains more than 2% of the vertices in the original graph). Further work by Karypis and Kumar accelerates the running time of multilevel recursive bisection by developing a multilevel p-way partitioning algorithm in which coarsening and refinement are performed a single time rather than at every step of the bisection.

Recent progress in graph partitioning has been very impressive. Large graphs are being partitioned faster and better than ever before. Graphs with millions of edges can be partitioned well in seconds.

Future work would be well-served by an evaluation of the partitioning alternatives which would also be very interesting. There are many packages available. Jostle supports parallel partitioning, with Walshaw's emphasis on minimizing vertex movement [24]. Chaco [23] is Hendrickson and Leland's package that contains implementations of their 4-way/8-way spectral algorithms, multilevel algorithms, and refinements to KL/FM. Other online packages worth investigating are Scotch [22] and Party [21]. These partitioning alternatives will result in new partition to be checked within the Virtex2 device and may give better results than the ones we presented here so that the partitions can be further used to improve the work done by Sourdis and Pnevmatikatos.

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- [29] SNORT official web site. <u>http://www.snort.org</u>.
- [30] METIS http://www-users.cs.umn.edu/~karypis/metis/
Appendix A

Here we include the results for our partitions on Virtex2 machine

Sourdis 6-partitions

Design Summary				
Number of errors: 0				
Number of warnings: 1				
Logic Utilization:				
Number of Slice Flip Flops:	15,677	out of	28,672	54%
Number of 4 input LUTs:	9,637	out of	28,672	33%
Logic Distribution:				
Number of occupied Slices:	8,567	out of	14,336	59%
Total Number 4 input LUTs:	13,946	out of	28,672	48%
Number used as logic:	9,637			
Number used as Shift registers:	4,309			
Number of bonded IOBs:	32	out of	484	6%
Number of GCLKs:	2	out of	16	12%

Total equivalent gate count for design: 459,020 Additional JTAG gate count for IOBs: 1,536 Peak Memory Usage: 567 MB

Asterisk (*) preceding a constraint indicates it was not met. This may be due to a setup or hold violation.

Constraint	Requested	Actual	Logic
			Levels
TS_clock_f = PERIOD TIMEGRP "clock_f"	3 3.000ns	2.812ns	1
nS HIGH 50.000000 %			
TS_clock_s = PERIOD TIMEGRP "clock_s"	6 6.000ns	3.014ns	0
nS HIGH 50.000000 %			

```
All constraints were met.
Section 12 - Configuration String Details
BUFGMUX "clock f ibuf/BUFG": Configuration String is:
  "IO_USED:0 SINV:S_B DISABLE_ATTR:LOW"
BUFGMUX "clock_s_ibuf/BUFG": Configuration String is:
   "IO_USED:0 SINV:S_B DISABLE_ATTR:LOW"
Section 13 - Additional Device Resource Counts
-----
Number of JTAG Gates for IOBs = 32
Number of Equivalent Gates for Design = 459,020
Number of RPM Macros = 0
Number of Hard Macros = 0
CAPTUREs = 0
BSCANs = 0
STARTUPs = 0
PCILOGICs = 0
DCMs = 0
GCLKs = 2
ICAPs = 0
18X18 Multipliers = 0
Block RAMs = 0
TBUFs = 0
Total Registers (Flops & Latches in Slices & IOBs) not driven by LUTs = 2807
IOB Dual-Rate Flops not driven by LUTs = 0
IOB Dual-Rate Flops = 0
IOB Slave Pads = 0
IOB Master Pads = 0
IOB Latches not driven by LUTs = 0
IOB Latches = 0
IOB Flip Flops not driven by LUTs = 0
IOB Flip Flops = 0
Unbonded IOBs = 0
Bonded IOBs = 32
Total Shift Registers = 4309
Static Shift Registers = 0
Dynamic Shift Registers = 4309
16 \times 1 \text{ ROMs} = 0
16x1 \text{ RAMs} = 0
32x1 \text{ RAMs} = 0
Dual Port RAMs = 0
MUXFs = 0
MULT ANDs = 0
4 input LUTs used as Route-Thrus = 0
4 input LUTs = 9637
Slice Latches not driven by LUTs = 0
Slice Latches = 0
Slice Flip Flops not driven by LUTs = 2807
Slice Flip Flops = 15677
Slices = 8567
Number of LUT signals with 4 loads = 0
Number of LUT signals with 3 loads = 0
Number of LUT signals with 2 loads = 3
Number of LUT signals with 1 load = 8555
NGM Average famout of LUT = 2.68
NGM Maximum fanout of LUT = 16
NGM Average famin for LUT = 3.1479
Number of LUT symbols = 9637
Number of IPAD symbols = 20
Number of IBUF symbols = 20
Number of LUTs replicated during cover = 951
```

Our 6-partition

Design Summary				
Number of errors: 0				
Number of warnings: 1				
Logic Utilization:				
Number of Slice Flip Flops:	15,552	out of	28,672	54%
Number of 4 input LUTs:	9,598	out of	28,672	33%
Logic Distribution:				
Number of occupied Slices:	8,616	out of	14,336	60%
Total Number 4 input LUTs:	13,816	out of	28,672	48%
Number used as logic:	9,598			
Number used as Shift registers	3: 4,218			
Worken of bould IODer			10.4	<i>~</i> •
Number of bonded 1083:	34	out or	484	0%
Number of GCLKs:	2	out of	16	12%
Total equivalent gate count for Additional JTAG gate count for 1 Peak Memory Usage: 554 MB	design: 4 IOBs: 1,53	451,962 36		

Constraint	Requested	Actual	Logic
			Levels
TS_clock_f = PERIOD TIMEGRP "clock_f" :	3 3.000ns	2.732ns	1
nS HIGH 50.000000 %			
TS_clock_s = PERIOD TIMEGRP "clock_s"	5 6.000ns	2.767ns	0
nS HIGH 50.000000 %			

All constraints were met.

```
Section 13 - Additional Device Resource Counts
      _____
Number of JTAG Gates for IOBs = 32
Number of Equivalent Gates for Design = 451,962
Number of RPM Macros = 0
Number of Hard Macros = 0
CAPTUREs = 0
BSCANs = 0
STARTUPs = 0
PCILOGICs = 0
DCMs = 0
GCLKs = 2
ICAPs = 0
18X18 Multipliers = 0
Block RAMs = 0
TBUFs = 0
Total Registers (Flops & Latches in Slices & IOBs) not driven by LUTs = 2789
IOB Dual-Rate Flops not driven by LUTs = 0
IOB Dual-Rate Flops = 0
IOB Slave Pads = 0
IOB Master Pads = 0
IOB Latches not driven by LUTs = 0
IOB Latches = 0
IOB Flip Flops not driven by LUTs = 0
IOB Flip Flops = 0
Unbonded IOBs = 0
Bonded IOBs = 32
Total Shift Registers = 4218
Static Shift Registers = 0
Dynamic Shift Registers = 4218
16x1 \text{ ROMs} = 0
16x1 \text{ RAMs} = 0
32x1 \text{ RAMs} = 0
Dual Port RAMs = 0
MUXFs = 0
MULT ANDs = 0
4 input LUTs used as Route-Thrus = 0
4 input LUTs used as Route-Thrus = 0
4 input LUTs = 9598
Slice Latches not driven by LUTs = 0
Slice Latches = 0
Slice Flip Flops not driven by LUTs = 2789
Slice Flip Flops = 15552
Slices = 8616
Number of LUT signals with 4 loads = 0
Number of LUT signals with 3 loads = 1
Number of LUT signals with 2 loads = 1
Number of LUT signals with 1 load = 8540
NGM Average fanout of LUT = 2.65
NGM Maximum fanout of LUT = 16
NGM Average famin for LUT = 3.1399
Number of LUT symbols = 9598
Number of IPAD symbols = 20
Number of IBUF symbols = 20
Number of LUTs replicated during cover = 996
```

Sourdis 12-partition

Design Summary			
Number of errors: 0			
Number of warnings: 1			
Logic Utilization:			
Number of Slice Flip Flops:	17,866 out of	21,504	83%
Number of 4 input LUTs:	10,344 out of	21,504	48%
Logic Distribution:			
Number of occupied Slices:	9,577 out of	10,752	89%
Total Number 4 input LUTs:	15,735 out of	21,504	73%
Number used as logic: 10,344			
Number used as Shift registers: 5,391			
Number of bonded IOBs:	32 out of	624	5%
Number of GCLKs:	2 out of	16	12%
Total equivalent gate count for design: 550,022			

Additional JTAG gate count for IOBs: 1,536 Peak Memory Usage: 560 MB

Constraint	Requested	Actual	Logic
			Levels
TS_clock_f = PERIOD TIMEGRP "clock_f"	3. 3.800ns	3.149ns	1
800 nS HIGH 50.0000000 %			
TS_clock_s = PERIOD TIMEGRP "clock_s"	7. 7.600ns	2.451ns	0
600 nS HIGH 50.0000000 %			

All constraints were met.

```
Section 13 - Additional Device Resource Counts
 -----
Number of JTAG Gates for IOBs = 32
Number of Equivalent Gates for Design = 550,022
Number of RPM Macros = 0
Number of Hard Macros = 0
CAPTUREs = 0
BSCANs = 0
STARTUPs = 0
PCILOGICs = 0
DCMs = 0
GCLKs = 2
ICAPs = 0
18X18 Multipliers = 0
Block RAMs = 0
TBUFs = 0
Total Registers (Flops & Latches in Slices & IOBs) not driven by LUTs = 3480
IOB Dual-Rate Flops not driven by LUTs = 0
IOB Dual-Rate Flops = 0
IOB Slave Pads = 0
IOB Master Pads = 0
IOB Latches not driven by LUTs = 0
IOB Latches = 0
IOB Flip Flops not driven by LUTs = 0
IOB Flip Flops = 0
Unbonded IOBs = 0
Bonded IOBs = 32
Total Shift Registers = 5391
Static Shift Registers = 0
Dynamic Shift Registers = 5391
16 \times 1 \text{ ROMs} = 0
16x1 \text{ RAMs} = 0
32x1 \text{ RAMs} = 0
Dual Port RAMs = 0
MUXFs = 0
MULT AND s = 0
4 input LUTs used as Route-Thrus = 0
4 input LUTs = 10344
Slice Latches not driven by LUTs = 0
Slice Flip Flops not driven by LUTs = 3480
Slice Flip Flops = 17866
Slices = 9577
Number of LUT signals with 4 loads = 0
Number of LUT signals with 3 loads = 0
Number of LUT signals with 2 loads = 1
Number of LUT signals with 1 load = 8993
NGM Average fanout of LUT = 2.96
NGM Maximum fanout of LUT = 16
NGM Average fanin for LUT = 3.0823
Number of LUT symbols = 10344
Number of IPAD symbols = 20
Number of IBUF symbols = 20
```

```
Number of LUTs replicated during cover = 721
```

Our 12-partition

Design Summary		
Number of errors: O Number of warnings: 1 Logic Utilization:		
Number of Slice Flip Flops: Number of 4 input LUTs:	17,808 out of 21,504 82% 10,527 out of 21,504 48%	
Logic Distribution: Number of occupied Slices:	9 458 out of 10 752 87%	
Number 4 input LUTs: Number used as logic: Number used as Shift registers:	15,828 out of 21,504 73% 10,527 5,301	
Number of bonded IOBs: Number of GCLKs:	32 out of 624 5% 2 out of 16 12%	
Total equivalent gate count for de Additional JTAG gate count for IOB Peak Memory Usage: 560 MB	sign: 544,896 3s: 1,536	
Constraint	Requested Actual 	Logic Levels
TS_clock_f = PERIOD TIMEGRP 800 nS HIGH 50.000000 %	"clock_f" 3. 3.800ns 3.219ns 	1
TS_clock_s = PERIOD TIMEGRP 600 nS HIGH 50.0000000 %	"clock_s" 7. 7.600ns 2.205ns 	0
Number of Equivalent Gates for De Number of RPM Macros = 0 Number of Hard Macros = 0 CAPTUREs = 0 BSCANS = 0 STARTUPS = 0 PCILOGICS = 0 DCMS = 0 GCLKS = 2 ICAPS = 0 18X18 Multipliers = 0 Block RAMS = 0 TBUFS = 0 Total Registers (Flops & Latches IOB Dual-Rate Flops not driven by IOB Dual-Rate Flops = 0 IOB Slave Pads = 0 IOB Slave Pads = 0 IOB Latches not driven by LUTs = IOB Latches = 0 IOB Flip Flops not driven by LUTs IOB Flip Flops not driven by LUTs IOB Flip Flops = 0 Unbonded IOBS = 32 Total Shift Registers = 5301 Static Shift Registers = 5301 I6X1 ROMS = 0 16X1 RAMS = 0 Jual Port RAMS = 0 MUXFS = 0	<pre>esign = 544,896 in Slices & IOBs) not driven by LUTs = 3305 y LUTs = 0 0 s = 0</pre>	

```
Slice Latches not driven by LUTs = 0
Slice Latches = 0
Slice Flip Flops not driven by LUTs = 3305
Slice Flip Flops = 17808
Slices = 9458
Number of LUT signals with 4 loads = 0
Number of LUT signals with 3 loads = 1
Number of LUT signals with 2 loads = 0
Number of LUT signals with 1 load = 9199
NGM Average fanout of LUT = 2.89
NGM Maximum fanout of LUT = 16
NGM Average famin for LUT = 3.0494
Number of LUT symbols = 10527
Number of IPAD symbols = 20
Number of IBUF symbols = 20
Number of LUTs replicated during cover = 805
```