

Automatic Structural Inference of Binary Protocols

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May 13, 2013

Abstract

Communication protocols are the foundation of everyday networking tasks but they are not always open in terms of available documentation. For reasons of interoperability or the collection of statistics some protocols need to be reverse engineered. This thesis presents a method based on density-based clustering and byte distribution classification for automatically inferring the structure of a protocol from a packet dump. We evaluate the precision of the method for finding different packet types and for finding fields, both globally across all packets and within the inferred packet types.

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Chapter 1

Introduction

1.1 Background

Communication protocols are the foundation of everyday networking tasks. Each time a web page is served or an email is sent, a rigorous series of actions is taken to ensure that the requested piece of information is relayed. The information is packaged in a *message* and both the syntax and the semantics of these messages are defined by protocols.

Protocols may be open or closed in terms of available documentation. The reason for them being closed might be that they are proprietary or simply because no one invested in creating a documentation of the protocol in the first place. Despite the lack of a documentation there still sometimes exists a need to understand the inner workings of a certain protocol, e.g. when providing interoperability between services. One infamous example of this is Microsoft's proprietary SMB protocol which spawned the Samba project. This project was created in an attempt to provide access to certain Windows services for Unix-like operating systems. Tridgell (2003) explains that the Samba project reversed the SMB specification using techniques like *fuzzing*, a process where they inspect the response from a SMB server for a wide variety of generated requests. Using this approach, it took many years until most of the functionality of the SMB protocol had been implemented.

Another area where the need for inferring the structures of closed protocols is in *deep packet inspection* (DPI). Messages sent over a network are usually constructed hierarchically. At the bottom there is only raw binary data and at the top there is the application data, e.g. an email or a video. There in between exists several layers that contain metadata which is used to make sure that the data reaches its target and that it does not get corrupted during transmission. As a message is sent from one party to another it traverses a series of nodes in the network. These nodes only need to look at the lower parts of the hierarchical structure of the message in order to relay it. If a node wants to conduct DPI however, the higher levels of the hierarchy are looked at as well. To be able to do this the structure of the protocol needs to be known.

There are many different reasons why one would want to perform DPI, one of which is to provide *quality of service* (QoS). QoS is needed in order to gather statistics for Internet service providers that manage the networks. One company

that delivers QoS solutions is Procera Networks on whose behalf we are writing this thesis.

In this thesis we have focused on finding the structure of binary protocols as opposed to textual ones. The reason for this being that textual protocols are intrinsically simple to read and understand and therefore do not require advanced methods of analysis. Some of the methods described here are however also applicable to textual protocols.

1.2 Related work

A popular approach to the problem of automatic inference of network protocols has been to draw a parallel to bioinformatics. Byte-streams have been processed using the same methodology as DNA sequence analysis in order to mine patterns in protocol messages. (Beddoe, 2005) explores the possibilities of using bioinformatics methods such as multiple sequence alignment in order to find patterns in both textual and binary protocols. These principles have been applied in e.g. Netzob¹, a tool for protocol analysis.

Using techniques from bioinformatics in protocol analysis seems natural since the analysis of messages, which may be viewed as the analysis of byte-streams, is easily converted to the problem of sequence analysis. Methods such as multiple sequence alignment does provide an insight to the structure of protocols, but further analysis is needed in order to find semantic meaning in segments of bytes.

Discoverer by Cui et al. is an approach which is specifically modeled after the behavior of network protocols. Discoverer tokenizes network dumps into textual and binary tokens. From the tokens Discoverer attempts to group messages based on their true message type, such as requests or responses. After the grouping of messages, Discoverer seeks semantic meaning in the tokens in common for each group of messages, and classifies the tokens into a set of predefined classes.

Caballero and Song (2012) proposes a method where protocol specifications are constructed by monitoring the execution state of the client or server software which utilizes the protocol. This approach has the obvious benefit of mapping data stored in memory at one state to the data subsequently present in a protocol message. This requires access to the actual software which generates the messages. In many cases this may limit the analysis to the messages sent by the client. When the server software is unavailable for execution state monitoring, one must resort to other methods in order to further analyze the protocol.

¹<http://www.netzob.org>

Chapter 2

Theory

2.1 Features

In machine learning, a feature is a measurable property of the input data. An example of a feature when working with images could be the intensity of light for a certain pixel. Features do not need to directly correspond to a physically measurable property however, instead they can be constructed by transforming other features. These new high-level features can then be used just like any other feature in order to simplify a complex problem.

One common type of complex problem is working with high dimensional data. There are different ways to deal with such data; one of them being to use a solving method that scales linearly. Another way is to reduce the dimensionality of the problem itself.

2.1.1 Principal component analysis

One method to reduce the dimensionality of a problem is to use principal component analysis (PCA). PCA is defined by Jolliffe (1986) and transforms the input data into another dataset where each dimension accounts for as much variance as possible while requiring it to maintain orthogonality to the other dimensions. The new dataset can be constructed to contain fewer dimensions than the original data and will then capture as much variance as possible in the given number of dimensions. PCA can be accomplished through singular value decomposition of a matrix.

2.2 Simple linear regression

In statistics, a response variable Y_i that is linearly dependent on an explanatory variable X_i can be modeled with a simple linear regression model. The relationship between the two variables are then

$$Y_i = \alpha + \beta X_i + \varepsilon_i \quad (2.1)$$

where α and β are the *regression coefficients* and ε_i is the error term. Given a set of data $\{Y_i, X_i\}_1^n$ the regression coefficients can be estimated through linear regression. This process is often called fitting as it can be seen as trying to find

Algorithm 1: RANSAC

Input: $D, M, n, \delta, \gamma, k$
Output: P_b

$P_b \leftarrow \emptyset$
 $\varepsilon_b \leftarrow \infty$
for k **do**
 $I \leftarrow n$ number of randomly chosen samples
 fit M to I
 $I_m \leftarrow$ samples where distance from $M \leq \delta$
 $I \leftarrow I \cup I_m$
 if $|I| \geq \gamma$ **then**
 fit M to I
 $\varepsilon \leftarrow$ deviation from M for all I
 if $\varepsilon < \varepsilon_b$ **then**
 $P_b \leftarrow$ parameters in M
 $\varepsilon_b \leftarrow \varepsilon$
return P_b

the model with the best fit for the data. There are different methods to do this as there is no consistent measure for what is the best fit.

2.2.1 Ordinary least squares

Ordinary least squares (OLS) is a method for estimating the regression coefficients in equation 2.1. It does this by minimizing the sum of squared residuals of the simple linear regression model. This can be formulated as minimizing the following cost function.

$$C(\alpha, \beta) = \sum_{i=1}^n (Y_i - \alpha - \beta X_i)^2 \quad (2.2)$$

The result of this is a model that minimizes the residuals globally across all samples. This intrinsically makes the method *non-robust*, i.e. sensitive to outliers.

2.2.2 RANSAC

Random Sample Consensus (RANSAC) (Fischler and Bolles, 1981) is a *robust* method for fitting a dataset to a model and was originally intended for image analysis. The robustness of the method allows it to avoid the influence of outliers in the model. Given a dataset D , a model M , a set of inliers I and two confidence parameters, δ and γ , it works the following way:

1. Fit M to I
2. Find the samples that are within δ distance from M and add those samples to I
3. Continue if $|I|$ is at least γ



Figure 2.1: A comparison between OLS and RANSAC on a dataset containing outliers.

4. Fit M to I
5. Evaluate M from the residual of I

One problem with this approach is that an initial set of inliers is required. In most cases this is not available so we need to guess which samples might be inliers. A simple solution to this is to take n random samples and assume that they are inliers. This only works if the assumption is actually true and thus the resulting algorithm will be *non-deterministic*, meaning that it only produces a good fit with a certain probability. The algorithm is normally run k number of iterations until the probability of success is sufficiently large. Pseudocode for RANSAC can be seen in algorithm 1.

The model that RANSAC operates on can be any kind of mathematical model, therefore when handling two-dimensional linear data the simple linear regression model given in equation 2.1 can be used. This gives us a method that solves the problem of simple linear regression but that is also insensitive to outliers. A comparison between OLS and RANSAC on a dataset where some samples have been replaced by random noise can be seen in figure 2.1.

2.3 Clustering

A clustering is a grouping of samples based on similarity with respect to their features. The features are predefined to represent distinct properties of the samples. Similarities between samples are normally represented as distances in an n -dimensional space, where n is the number of features. Distances may be calculated using any suitable norm.

In cluster analysis, the goal is normally to find a clustering from a set of samples such that the membership of a cluster represents some true relationship. In

some clustering algorithms such as K-means (MacQueen et al., 1967), the number of clusters are predefined. When the number of clusters are unknown, other algorithms such as DBSCAN, OPTICS and hierarchical clustering methods may be used.

The clustering algorithm we use in our method is OPTICS, which is based on DBSCAN. In the rest of this section we will explain these algorithms and provide examples based on protocol data.

2.3.1 DBSCAN

DBSCAN is a *density-based* clustering algorithm, as opposed to partitioning clustering algorithms such as K-means. Density-based clustering algorithms provides the benefit of not having to provide an estimated number of clusters as a parameter to the algorithm.

DBSCAN was first presented by Ester et al. (1996). The algorithm takes a set of samples D and two parameters, $MinPts$ and ε . $MinPts$ decides how many samples that are needed in order to form a cluster.

The densities are defined by the distances between a set of points. The algorithm introduces the definition ε -neighborhood $N_\varepsilon(p)$ for a point p . The samples which are in $N_\varepsilon(p)$ are given by the following condition:

$$N_\varepsilon(p) = \{q \in D \mid dist(p, q) < \varepsilon\} \quad (2.3)$$

The samples in $N_\varepsilon(p)$ are defined as *directly density-reachable* from p . Given that $|N_\varepsilon(p)| \geq MinPts$, the samples in $N_\varepsilon(p)$ forms a cluster, and p is a *core point*.

A cluster is not limited to containing samples which are directly density-reachable. The DBSCAN algorithm also defines that samples which are not directly density-reachable may be *density-reachable*. A sample q is density-reachable from a sample p if there is a set of samples $S = \{s_1, \dots, s_n\}$ where $p = s_1$, $q = s_n$ and s_{i+1} is directly density-reachable from s_i for $0 < i < n$.

Since the density-reachable relationship is not symmetric, a looser relationship is introduced: *density-connected*. Two samples p and q are density-connected if there exists a third point o from which both p and q are density-reachable. The density-connected relationship is symmetric.

With these relationships, the algorithm defines cluster membership as follows:

Definition. *The following conditions needs to be satisfied for a point q to be a member of a cluster C , given that there is a sample $p \in C$:*

1. *q is density-reachable from p*
2. *q is density-connected to p*

One of the drawbacks of DBSCAN is that it can only find clusters with a density higher than the density decided by the ε -parameter. It is also hard to estimate the ε and $MinPts$ -parameters for an unknown dataset. This makes DBSCAN suitable for classifying data into known classes, but not as suitable for finding underlying structures in entirely unknown datasets. In figure 2.2 is an example of DBSCAN running on 5000 packets of DNS data. The parameters are manually selected to $\varepsilon = 0.085$ and $MinPts = 200$.



Figure 2.2: An example of clustering with DBSCAN running on samples generated from 5000 packets of DNS data. The samples has been projected down to two dimensions using PCA (see section 2.1.1).

2.3.2 OPTICS

Addressing the difficulties in selecting parameters for DBSCAN, Ankerst et al. (1999) introduced the OPTICS algorithm, which is an extension of the concepts introduced in DBSCAN. OPTICS uses the definitions *directly density-reachable*, *density-reachable* and *density-connected* which were described in the DBSCAN algorithm, but eliminates the need for an explicit ε parameter. Instead the ε parameter is interpreted as the largest distance to consider when clustering. The results is an algorithm which is less sensitive to user-specified parameters, and is able to find clusters with varying densities.

One important aspect of OPTICS is that it does not generate actual clusters. Instead, it generates an *ordering* of samples and a set of corresponding *reachability-distances* which reveals density-based structure in the input data. The OPTICS algorithm extends DBSCAN with the definitions *core-distance* and *reachability-distance*.

Core-distance is a measurement of the distance ε which is required for a sample p to be a core point. That is, a core distance is the distance ε which satisfies $|N_\varepsilon(p)| = \text{MinPts}$.

Definition. The core-distance for a sample p is defined as:

$$\begin{cases} \text{UNDEFINED}, & \text{if } |N_\varepsilon(p)| < \text{MinPts} \\ \text{Min } \varepsilon \text{ which satisfies } |N_\varepsilon(p)| = \text{MinPts}(p), & \text{otherwise} \end{cases}$$

The core-distance is *UNDEFINED* when an upper limit on ε is given as a parameter to the algorithm. This parameter is not required, although it has an impact on the runtime of the algorithm.



(a) Reachability-plot for 5000 DHCP-packets with $MinPts = 200$. (b) Hierarchical extraction for 5000 DNS-packets with $MinPts = 200$.

Figure 2.3: Reachability-plots generated from the ordering and reachability-distances given by OPTICS.

The reachability-distance of a sample p is the smallest distance which is needed for p to be directly density-reachable to a core point q for some ε .

Definition. The reachability-distance of a sample p with respect to some sample q is defined as:

$$\begin{cases} UNDEFINED, & \text{if } |N_\varepsilon(q)| < MinPts \\ \max(core-distance(q), distance(q, p)), & \text{otherwise} \end{cases}$$

where $distance(q, p)$ is the smallest ε for which q is density-reachable from p .

From the ordering and reachability-distances, the cluster structure may be visualized in a *reachability-plot*. The reachability-plot provides an overview of the output from OPTICS as a bar-plot where the bars represent reachability-distances in the ordering generated by OPTICS.

An example of a reachability plot generated from 5000 DHCP-packets with $MinPts = 500$ is shown in figure 2.3a. Valleys in the reachability-plot represents samples which are close to each other. Spikes represents a large difference in distance between the samples to the left and right of the spike.

The pseudocode algorithm is given in algorithm 2.

Algorithm 2: OPTICS

```
Input:  $D, MinPts, \varepsilon$   
Output: ordering, reachability-distances  
reachability-distances  $\leftarrow \emptyset$   
ordering  $\leftarrow \emptyset$   
seeds  $\leftarrow 0, 1, \dots, |D|$   
 $i \leftarrow 0$   
while  $|seeds| > 1$  do  
   $p \leftarrow seeds.get(i)$   
   $seeds.remove(i)$   
  ordering.add( $p$ )  
  if  $core-distance_{\varepsilon, MinPts}(p) \leq \varepsilon$  then  
    neighbors  $\leftarrow N_{\varepsilon}(p)$   
    for each neighbor sample  $n$  do  
       $rdist \leftarrow reachability-distance_{\varepsilon, MinPts}(n, p)$   
      reachability-distances[ $n$ ]  $\leftarrow rdist$   
     $i \leftarrow$  seed with least reachability-distance  
  else  
     $i \leftarrow seeds.first$   
/* Process last remaining seed */  
ordering.add(seeds.first)  
reachability-distances[0]  $\leftarrow 0$   
return ordering, reachability-distances
```

Cluster extraction

Ankerst et al. describes the principles of the OPTICS algorithm as running DBSCAN for an infinite number of distance parameters ε_i in the interval $0 \leq \varepsilon_i \leq \varepsilon$. They also explain that extracting clusters from the OPTICS ordering and reachability-distances using a static reachability-distance threshold ε_i gives a clustering which is roughly equivalent to the clustering obtained when running DBSCAN with the same *MinPts* and $\varepsilon = \varepsilon_i$.

Sander et al. (2003) describes a hierarchical cluster extraction algorithm for OPTICS. As opposed to the DBSCAN-equivalent extraction approach, hierarchical cluster extraction retains many of the properties which comes with OPTICS, such as finding clusters with varying densities and subclusters nested in larger clusters.

The algorithm takes an ordering and reachability-distances from OPTICS and builds a hierarchical representation of the clustering in the form of a tree, where the leaves are clusters.

The pseudocode for hierarchical extraction is given in algorithm 3. Figure 2.3b is the result of the algorithm running on the same DNS dataset as figure 2.2 with *MinPts* = 200. The algorithm yields three clusters, each cluster distinguished by changing background color. The true types *request* and *response* are colored blue and green respectively in the reachability plot. Note that the algorithm is rather insensitive to the *MinPts*-parameter, and that the true types of DNS may be extended to different record types such as A/AAAA/CNAME-records.

2.3.3 Clustering metrics

When measuring the quality of a clustering from a clustering algorithm, a number of metrics are defined. These metrics reflect the quality of the features as well as the clustering algorithm. The metrics require labels given from the clustering algorithm as well as a *truth*, i.e. the true classes of the samples. The clustering metrics in this section are defined by Rosenberg and Hirschberg (2007) using information entropy.

Homogeneity score

Homogeneity is satisfied when every cluster only contains members of a single true class. When completely satisfied, $h = 1$. If every cluster contains samples from different true classes then $h = 0$. When some clusters only contain members from one true class while others contain mixed true classes then $0 < h < 1$.

Completeness score

Completeness is satisfied when all members of a true class are in the same cluster. When completely satisfied, $c = 1$. If all members of a true class are spread out over different clusters then $c = 0$. When some true classes are spread out while others are contained in single clusters then $0 < c < 1$.

V-Measure score

V-Measure is a weighted harmonic mean of homogeneity and completeness, and is defined by Rosenberg and Hirschberg as:

$$V_{\beta} = \frac{(1 + \beta) * h * c}{(\beta * h) + c}$$

We weigh homogeneity and completeness equally in our application, thus setting $\beta = 1$.

2.4 Sequence alignment

The problem of aligning sequences with one another originally arose in the area of bioinformatics. There, a need to find similarities in long chains of amino acids existed. One of the first methods that solved this problem was the Needleman-Wunsch algorithm (Needleman and Wunsch, 1970). The algorithm finds the maximal number of matching symbols between two sequences, allowing for gaps to be inserted into either sequence. This is done by tracing a path through an alignment matrix where each element represents a possible matching of symbols between the sequences.

Originally, the path was built only with respect to the number of matching identical symbols and not to the number of mismatches or gaps. This has since been improved upon to include pairwise scores for symbol matching and penalties for gaps. The different scores used when matching symbols is typically given in a similarity matrix. An example of such a matrix for a small alphabet of symbols is seen in figure 2.4a.

Given two sequences, S and T , of length m and n respectively, a scoring matrix C and a gap penalty d the algorithm does the following:



Figure 2.4: An example of the Needleman-Wunsch algorithm run on two sequences constructed from the alphabet $\{A, C, G, T\}$ with a gap penalty $d = -1$. The similarity matrix used can be seen in (a) and the resulting alignments is given by the gray elements in (b).

1. Construct an $(n + 1) \times (m + 1)$ alignment matrix A
2. Fill the first row of A with gap penalties: $a_{0,j} \leftarrow j \cdot d$ where $0 \leq j \leq m$
3. Fill the first column of A with gap penalties: $a_{i,0} \leftarrow i \cdot d$ where $0 \leq i \leq n$
4. Fill the rest of A by doing one of the following actions for each element $a_{i,j}$:
 - Match: $a_{i,j} \leftarrow a_{i-1,j-1} + C_{S_j,T_i}$
 - Delete: $a_{i,j} \leftarrow a_{i-1,j} + d$
 - Insert: $a_{i,j} \leftarrow a_{i,j-1} + d$

The action that is chosen is the one that results in the largest score where, in the case of a tie, match is given priority. The priority of delete and insert can be chosen arbitrarily.

5. Backtrack through the matrix from $a_{n,m}$ to build the alignments in reverse order. Choose the path that was used to construct the score at the current element. The algorithm finishes when element $a_{0,0}$ is reached.

An example of applying the algorithm can be seen in figure 2.4 and the resulting alignments in figure 2.5.

A-TTCGCT
ACTT-GC-

Figure 2.5: The resulting alignment of the sequences ATTCGCT and ACTTGC .

Algorithm 3: Hierarchical Cluster Extraction

Input: *ordering, reachability-distances*

Output: *clustering*

$R \leftarrow$ *reachability-distances* arranged according to *ordering*

$n \leftarrow$ number of samples

$L \leftarrow$ indices of local maxima in R

Sort L on $R[L_i]$

$R_{max} \leftarrow R[L.last]$

$leaves \leftarrow \emptyset$

function *cluster_tree*(*node, parent, L*):

if L is empty **then**

$leaves.add(node)$

return

$s \leftarrow L.pop()$

$node.split \leftarrow s$

$sign_thres \leftarrow significant_ratio * R[s]$

 Create two new nodes, $N1$ and $N2$

$N1.samples \leftarrow$ samples left of s

$N2.samples \leftarrow$ samples right of s

$L1 \leftarrow$ local maxima left of s

$L2 \leftarrow$ local maxima right of s

if $N1$ and $N2$ has average reachability $< sign_thres$ **then**

if $|N1| > MinPts$ **then**

$children.add(\{N1, L1\})$

if $|N2| > MinPts$ **then**

$children.add(\{N2, L2\})$

if $children$ is empty **then**

$leaves.add(node)$

return

if $R[s] \approx R[parent.split]$ **then**

for $\{child, L\}$ in $children$ **do**

$parent.children.add(child)$

$parent.children.remove(node)$

$p \leftarrow parent$

else

for $\{child, L\}$ in $children$ **do**

$node.children.add(child)$

$p \leftarrow node$

for $\{child, L\}$ in $children$ **do**

$cluster_tree(child, p, L)$

else

$cluster_tree(node, parent, L)$

Create node *root* containing all samples

$cluster_tree(root, null, L)$

Chapter 3

Method

3.1 Approach

A protocol is generally made up of a predefined set of message types. A message type may define if the message is a request or response. If so, it is fairly easy to determine which parts of the message that defines if it is a request or a response. One may simply look for values in a message that depends on the direction of the message. However, most protocols are not limited to being classified as requests and responses. Protocols often define sets of control message types which changes the state of the protocol. Some protocols does not follow the request/response model, which is the case for many peer-to-peer protocols.

In order to thoroughly analyze a protocol, the first step is essentially to distinguish the different types of the protocol. This is a necessary first step which enables us to infer the different states of the protocol and allows for type-specific field analysis. The type is commonly specified by some flag or a combination of flags. A type flag may be a byte, a bit subset of a byte, or spread out over multiple bytes where each byte may define some subtype to a more general type.

Messages that have the same value in their type flags are likely to have a distinguishable structure which differs from messages with other type flags. Some message types may introduce fields which are not present in other message types, and the fields which are global for the protocol are more likely to be identical within a type. A message that is used for binary data transmission is often longer than a control message.

We group messages that are similar using cluster analysis. We then move on to analyzing the grouped messages in an effort to discover the bytes which we rank the most probable to be responsible to the grouping we have found, and label these as (need new term here) the bytes which distinguishes a specific type. Once these bytes have been discovered, we may regroup the messages based on the value of that the byte takes.

From the new grouping, we are able to analyze the type specific properties of messages of a certain type. We focus on determining the field structure of each type, and make an effort to determine the semantics of each field. From the byte value distribution at a certain byte range, we try to classify each field as one of our primitive field types: constant, flag, uniform, number, incremental or

length. We do this analysis in global scope, cluster scope, stream and connection scope. We define the global scope to cover the fields that all messages have in common e.g. the protocol header. The cluster scope cover fields that are specific for a type. The connection scope is a subscope to the global and cluster scopes, and cover fields that are constant or incremental within a connection, the same goes for the stream scope but only in one direction.

3.2 Type inference

When inferring the message types, we perform a two-pass clustering. In the first pass, we use a clustering algorithm in order to statistically group messages that are similar according to the features we define. In the second pass, we utilize the assumption that there exists one or more bytes which distinguishes a specific type.

3.2.1 Initial clustering

The first step when performing cluster analysis is to define a set of features. The features which we have chosen are based on the byte value distribution at each offset for every message.

We define a matrix P where each element $P_{i,j}$ is the probability that the byte with the decimal value j is present at the offset i for all packets. Using P we create the feature vector $\{f_1, f_2, \dots, f_n\}$ for every message m where m is n bytes long. We define each feature as $f_i = P_{i,m_i}$.

Informally, this means that we create a feature vector consisting of n features for each message. Each element f_i in the feature vector is the probability that the byte value m_i occurs at offset i .

The reason for having probability-based features is that density-based clustering algorithms such as OPTICS rely on distance measures. If we were to take the actual byte values at each offset as features, the difference between two byte values would represent a distance between the samples. In a flag byte this would introduce a great distance between a message where the most significant bit is set and when it is not set. Since nominal features are not an option due to difficulties with high dimensionality, the probability measure is a compromise to having nominal features.

Once the features have been calculated, we apply PCA transformation parametrized to capture 80% of the variance. This reduces the dimensionality of our feature vectors while conserving the most significant variance. We use these new feature vectors as input to OPTICS.

OPTICS works well for our application since it is insensitive to the parameters selected. We require an estimate *MinPts* parameter. The distance parameter has the default value $\varepsilon = \infty$.

The resulting clustering from OPTICS may have far more clusters than the number of real types in the protocol. This is expected, since messages that are identical on large partitions of the data will induce a low distance when running the OPTICS algorithm, which in its turn will result in high-density areas within the real types. Having more clusters than the number of real types are not a problem however. In the next step of the clustering, we make use of the fact that

the clusters are very homogenous, i.e. each cluster mainly contains messages of one real type.

3.2.2 Type distinguishers

Once we have obtained the OPTICS clustering our goal is to reduce the number of clusters to match the actual number of real types. We do this under the assumption that there is some byte or a number of bytes that define the type of each message. We call these bytes *type distinguishers*.

Many protocols define an explicit type flag in the form of a single byte. The type distinguishers which fit our definition may consist of a number of bytes. When combined, these bytes give a reasonable indication of what messages with high similarity have in common. In order to find these bytes we use an approach similar to the recursive clustering described by Cui et al. in Discoverer. Our approach is as follows:

1. For every byte offset from the beginning of all messages up to a certain limit we calculate how many values the byte takes. This is equivalent to how many real types that would be present if the chosen byte was a real type flag.

We introduce two criteria when considering a byte as a type distinguisher. The first criterion is *MaxNumTypes* which decides the maximum number of types the protocol is assumed to have. The second criterion is *MaxTypeRatio* which decides the maximum ratio between the number of packets with a certain value for the considered byte and the total number of packets. If $MaxTypeRatio = 0.6$, it prevents a byte from being considered a possible type distinguisher if it assumes the same value in more than 60% of all messages.

2. Now that we have the possible type distinguishers we rank the type distinguishers based on completeness score. For each possible type distinguisher byte we group the messages on every value which the byte takes. This gives us a new clustering. We measure the quality of each type distinguisher byte by comparing its clustering to the OPTICS clustering. This is achieved using completeness score with the OPTICS clustering as our truth.
3. Once our type distinguisher bytes have been scored, we take the bytes with the highest completeness score. For every byte we take, we calculate how many clusters this would yield if we were to cluster all messages on the distinct values that the combination of bytes takes. This process goes on until we reach the point that we can take no more type distinguisher bytes without exceeding *MaxNumTypes*.

3.3 Field analysis

With our type inference complete we now have a number of clusters that contain similar messages. The next step in inferring the structure of the protocol is to identify field boundaries and their contents. This is a very complex problem as there can be as many fields as there are bits in the messages and the boundaries

		A	T	T	C	G	C	T
	0	-1	-2	-3	-4	-5	-6	-7
A		2	1	0	-1	-2	-3	-4
C			1	0	2	1	0	-1
T				3	2	3	2	2
T					2	3	2	4
G						4	4	3
C							6	5

(a) The alignment matrix for sequences using non-mutual sequence alignment.

ATTCGCT
ACTTGC-

(b) The resulting alignment of the sequences.

Figure 3.1: An example of our non-mutual sequence alignment algorithm run on the two sequences **ATTCGCT** and **ACTTGC** using the same parameters as in figure 2.4.

may lie in between any pair of adjacent bits. The number of possible combinations of fields can therefore easily become too large to handle. Fortunately, most protocols are not designed to utilize each and every bit and instead focus on being extensible, thus there are some common design principles that may be exploited.

3.3.1 Simplifications

First of all we restrict where boundaries may lie. We require fields to be n bytes in length where n is a power of two and also that they must be aligned. We use the definition of *n-byte alignment* which means that the offset at which the data is located must be a multiple of n . That means that a four byte field may be located at offset 0, 4 or 12 but not at offset 5 or 10. Most protocols follow these principle for optimization reasons as a received packet will reside in a byte buffer in memory and accessing misaligned data can be expensive; see e.g. Assembly/Compiler Coding Rule 46 in Intel[®] 64 and IA-32 Architectures Optimization Reference Manual.

Another principle that is often used is that messages start with a *header*, a collection of fields of fixed length that is located before the data part of the message. Most protocols need a header if they send messages of different types or messages with variable length, or both. In those cases the header will include at least one type flag or length field respectively. A type flag can be a single bit or a collection of bits that indicate what type the current message is so that it can be parsed correctly. A length field is a numerical field that contains metadata on how long a part of the message is, also for parsing purposes. As the header itself usually does not contain any variable length fields no preprocessing is needed to identify them, however for the rest of the message, which is usually type specific, the data needs to be aligned.

3.3.2 Non-mutual sequence alignment

For our purposes we have introduced the term *mutual* sequence alignment to refer to traditional sequence alignment via the Needleman-Wunsch algorithm. Furthermore, we have developed a modified version of the Needleman-Wunsch algorithm that performs *non-mutual* sequence alignment. The difference be-

tween the two algorithms is that the non-mutual version only allows gap insertion into one of the two sequences. This restriction only works when the gap insertion is allowed for shorter one of the two sequences; otherwise the algorithm will never be able to reach the top left element during the backtracking. For simplicity let us assume that the longer sequence will always be supplied in S , therefore gaps may only be inserted into T . The non-mutual property is achieved by the following modifications:

1. After the first row has been filled with gap penalties, fill the diagonal with matches: $a_{j,j} \leftarrow a_{j-1,j-1} + C_{S_j,T_j}$ where $1 \leq j \leq n$
2. Do not fill the first column with gap penalties.
3. During the filling phase, only allow the match and insert actions, i.e. remove the delete action, and only fill the upper part of the matrix (above the diagonal).

If we run this algorithm on the same example as in figure 2.4 we get the result shown in figure 3.1..

3.3.3 Preprocessing

Using this method of non-mutual sequence alignment we align each message within each cluster with the longest message in that cluster. This way we get an effective method for doing multiple sequence alignment of all messages in a cluster in a linear number of alignments. To model that variable length data pushes the next field boundary forward we also enforce gap insertion from the right instead of from the left by simply reversing the messages before aligning them and then reversing them back again after the alignment is complete. During the alignment we also use a somewhat complex 256×256 scoring matrix to get a good result. It is constructed to provide the following scores:

- Identical byte values: 2
- Alphanumeric ASCII characters: 2
- Non-alphanumeric ASCII characters: 1
- Numerical distance ≤ 10 : 1
- Numerical distance ≤ 20 : 0
- Anything else: -1

In addition to aligning the messages for the cluster scope of the analysis we also group messages by stream and connection. These groups are then used when attempting to find fields that are constant in the stream or connection scope. This is accomplished by looking at each message at the IP level and grouping them together depending on their combination of addresses and ports. After this preprocessing of the messages has been done we are now ready to start classifying fields.

3.3.4 Field classification

Now that we have limited our possible field locations and prepared our data for searching in different scopes, we try to classify our data into a predefined set of field types. This is done by iterating through all possible aligned locations of a set of field sizes, typically $\{2^n \mid 0 \leq n \leq 2\}$, and testing if the data at that location may belong to any of the predefined field types. We will then get a number of different possible types for each byte position, possibly of different sizes that might be overlapping. This is our estimate of the field types. To get the most likely candidate we can then give precedence to different sizes and types, favoring larger sizes and more complex types. The field types we have identified are the following:

- Constant
- Flag
- Uniform
- Number
- Incremental
- Length

These field types have been chosen because they are commonly used in the construction of protocols and because they exhibit some degree of identifiability. For the top three types we use simple techniques on the distributions of the individual bytes of all the messages in the current scope while for the bottom three we need to resort to more advanced analytical techniques on the data of each individual message.

3.3.5 Constant fields

We define a constant field as a field that only assumes a single value in all the messages in the current scope. By looking at the byte value probability distributions, we classify an offset i as constant if $P_{i,v} = 1$ holds for some v , $0 \leq v \leq 255$. $P_{i,v}$ is an element in the probability matrix P which is defined in section 3.2.1.

Constant fields may be interpreted as one of the following:

- A true protocol constant. These are fields that are defined to be constant in the protocol specification. An example of such a field is the protocol field in SMB, which is a four-byte constant field with the value `0xff534d42`.
- A reserved field. Reserved fields are specified in case there is need for future additional information in a fixed-sized header. Reserved fields are commonly set to zero.
- A field which is constant in the current scope. For example, if a dump only contains a single connection, then a session ID field may be interpreted as a constant.



Figure 3.2: Byte value distribution of the 6th byte in the DNS header. This byte is the least significant byte of the two-byte query count field. In our dump its value is constantly one.

3.3.6 Flag fields

A flag field is a field that defines some property of a message. Typically, it assumes a very limited number of values and may be used for representing a message type, an error code, a status code and so on.

Flag fields may be interpreted as real numbers where each number represents a numbered entity, such as a type identifier. It may also represent a set of binary values where each bit is a flag itself, and is interpreted using a bitmask. An example of such a field is the DNS flags field. The flag field in DNS is two bytes long, where the first bit of the field indicates if it is a query (0) or response (1).

The first byte of the DNS flag field is depicted in figure 3.3. The amount of queries and responses are relatively equally distributed. The bar at byte value 0 = 00000000 represent the queries. The bar at byte value 128 = 10000000 represent the responses.

When classifying flag fields, we constrain the amount of values that a certain byte may assume. We classify a field as a flag if the total amount of values the field takes is greater than one single value and less than some upper bound.

3.3.7 Uniform fields

A pattern we observed while inspecting the byte value distribution for various fields was that some fields were uniformly distributed. The real types of these fields are normally some randomly generated value, such as a generated session identifier or a nonce.

When classifying fields as uniform, we compare the distribution of the byte values to the uniform probability distribution for each byte. If we assume the observed byte value is a random variable X and the expected value the scalar x then the uniform probability distribution for a byte is



Figure 3.3: Byte value distribution for the 3rd byte in the DNS header. From this byte it may be determined if the message is a query or a response.

$$Pr[X = x] = \begin{cases} \frac{1}{256} & 0 \leq x \leq 255 \\ 0 & \text{otherwise} \end{cases}$$

If the sum of the deviations from this distribution is below a certain threshold then the byte is classified as being uniform. If all bytes are classified as uniform then the field is classified as uniform as well.

In figure 3.4 the value distribution for the first byte of the DNS header is displayed. This byte is the first byte of a two-byte identifier. This ID is generated by the client and used by the DNS server in order to identify to which query it issues a response. The black line represents the uniform distribution; a horizontal line at $\frac{1}{256}$.

3.3.8 Number fields

While every field in a binary protocol actually contains numbers, some exhibit a more natural distribution of these numbers than others. We define number fields as those that can be closely fitted to a normal distribution with constraints on the standard deviation. We set these constraints so that the distribution will not collide with our definitions for constant or uniform fields. This captures the behavior of a random variable X with mean μ and standard deviation σ . Typically, values that represent a count behave this way, e.g. the QDCOUNT or ANCOUNT fields of the DNS protocols that refer to how many questions or answers respectively the message contains. This behavior can also be seen in length fields as they represent a byte count. An example of such a byte count can be seen in figure 3.5.



Figure 3.4: Byte value distribution for the first byte of the DNS header. This byte is the first of a two-byte ID field.

3.3.9 Incremental fields

Some protocols contain fields where the value increases for each message in the scope. It can be a sequence number for each packet or a timestamp of when a certain action was performed. We define these as incremental fields. These can be identified by looking at all the messages within a scope and counting how many of them contain a value that is greater than the previous encountered value. We then compare this count to the total number of messages in the scope and if it is above a certain threshold we accept it as being incremental. The threshold value is used to allow for some amount of wrap-around of the values in the field.

3.3.10 Length fields

Length fields require the most analysis to identify but also gives a very confident estimation because of the depth of the analysis. As mentioned in section 3.3.1, length fields relate to how long a part of the message is. This part can be the entire message, the part following the header or just the next field. For a single message all these types of length fields can be found, as long as there is only one variable length part in the message. For the first case the length of the message will be a multiple of the numerical value in the field. Likewise, in the two last cases, the length will be a multiple of the numerical value in the field plus an offset equal to the length of the parts not covered by the field.

Unfortunately, if a message contains multiple parts of variable length the problem becomes more difficult and cannot be solved by looking at individual messages anymore. There is however a better way to approach this problem. We have already seen that the value of the length field multiplied by some scalar plus an offset (which may be zero) is equal to the length of the message. If x is

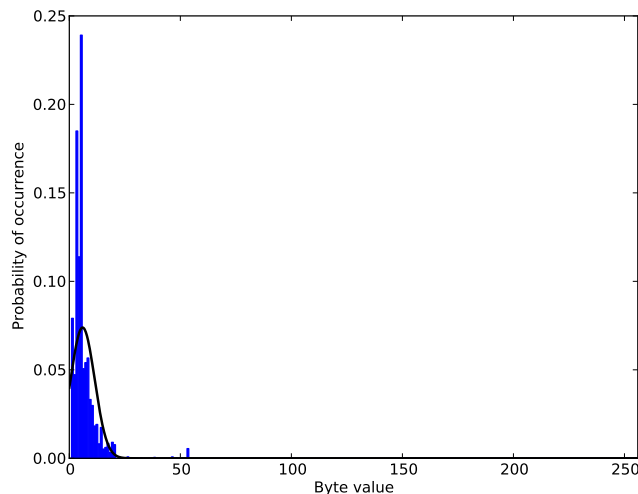


Figure 3.5: A fitted normal distribution to the distribution of values of the 13th byte in DNS.

the value of the field, y is the length of the message, α is the offset and β is the multiple then this can be written as

$$y = \alpha + \beta x$$

which means that this can be modeled with a linear model. We can then attempt to solve this with simple linear regression for several messages at once. If the samples are diverse enough there should be a linear relationship between the minimum size of all messages with a specific field value and the value itself. This represents the case of the other variable fields being minimized. The samples that are not of minimal size for the specific field value will then be seen as noise. Because of this noise, which normally is the dominant part of all samples, we need a robust method for fitting our linear model. We therefore use the RANSAC method as described in section 2.2.2 for this purpose. A real world example of this for the SMB protocol can be seen in figure 3.6.

One added advantage of using the RANSAC method is that it also handles other situations where noise appears. This can be if the protocol handles its own fragmentation or some other form of imperfect parsing that affects the length of the messages.

3.4 Protocol state inference

Many protocols may be described using a state diagram. Some handshake phase between the peers is common when a conversation is initialized. There might also be some teardown phase, and phases for transmitting data etc. Inferring the states of the protocol is an important task when trying to figure out the semantics of the protocol. Modelling the protocol states as a state diagram



Figure 3.6: RANSAC fit of the WC field in the SMB protocol. The resulting parameters are $\alpha = 39$ and $\beta = 2$.

provides a readable representation of the inner workings of the protocol, and might simplify the analysis of the fields of the different message types.

When a reasonable partitioning into types has been achieved, finding the protocol state diagram is trivial. We simply partition the set of messages into connections, where connections are tuples of client and server sockets. From each connection, we define each message in the connection as a change of state. Each state has a set of probabilities for transitioning to all other states we've defined. The cumulative sum of these probabilities sums up to 1.

We represent the states as two $n \times n$ -matrices, C and S , where n is the number of types, which will be the same as the number of states. Each entry $C_{i,j}$ or $S_{i,j}$ in the matrix defines the number of times we observe a transition by the client or server from state i to state j . We define the client as the peer initiating the connection. Once we have our two matrices, we may calculate the probabilities for the state transitions by dividing the total number of transitions between two states with the total number of messages.

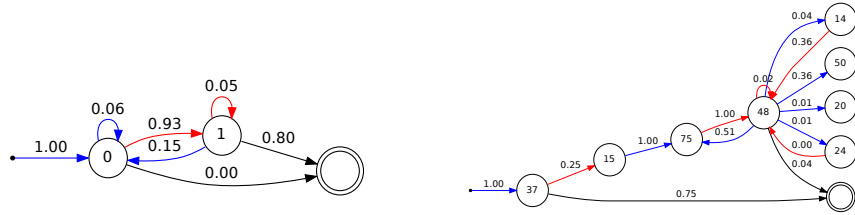
A state diagram of all protocol states and transitions may not always be feasible. Some protocols does not follow the state model which we're trying to model with our approach. They may allow some message types to be sent from all states, and may only be represented with a state diagram during some phases such as the handshake phase. We handle this problem by allowing the user to define the depth for which the analysis is performed. If depth is set to 5, we only analyse which states the protocol may be in after five transitions. From this restricted state diagram a handshake procedure will be much more readable.

In figure 3.8a the state diagram generated from a DNS stream capture with $MaxNumTypes = 3$ is shown. State 0 represents queries and state 1 represent responses. Blue edges represent messages sent by the client. Red edges represent messages sent by the client.

State	Type
37	Negotiate Protocol Request
15	Negotiate Protocol Response
75	Session Setup AndX Request
48	Session Setup AndX Response
14	Logoff AndX Request
50	Tree Connect AndX Request
20	NT Create AndX Request
24	Delete Request

Figure 3.7: SMB states.

Figure 3.8b shows the state diagram from SMB with a depth of five transitions. This shows the handshake used by the protocol. Table 3.7 below lists the true types of the messages in a left-to-right order with respect to the state diagram.



(a) State diagram for the DNS protocol. (b) State diagram for the SMB protocol

Figure 3.8: State diagrams generated by our state inference. The left-most dot represents the state before any communication has begun. The right-most double-circle represents the terminal state.

Chapter 4

Results

4.1 Data sets

Provide information about the data used to produce the results.

4.2 Clustering performance

Display the results of applying our method to different data sets.

4.3 Field inference performance

Protocol	# Messages	# Connections	# Types
DNS	30628	10935	6
SMB	27906	212	93
TFTP	6492	6	3
RTP	20179	5	1
DHCP	13014	376	11

Figure 4.1: Datasets.

Protocol	minSamples	maxNumTypes	Limit
DNS	200	10	100
SMB	50	150	100
TFTP	1	1	100
RTP	200	10	100
DHCP	1	1	100

Figure 4.2: Clustering parameters.

Protocol	Completeness	Homogeneity	V-measure
DNS	0.1642	0.9078	0.2781
SMB	0.8581	0.8559	0.8570
TFTP	1	1	1
RTP	0.0000	1.0000	0.0000
DHCP	1	1	1

Figure 4.3: Before type distinguisher clustering.

Protocol	Completeness	Homogeneity	V-measure
DNS	0.9534	0.9999	0.9761
SMB	0.9914	0.9880	0.9897
TFTP	1	1	1
RTP	0.0000	1.0000	0.0000
DHCP	1	1	1

Figure 4.4: After type distinguisher clustering.

Chapter 5

Discussion

5.1 Conclusions

Draw conclusions about our method and compare it to the related methods.

Is sequence alignment necessary? We could iteratively find length fields to get perfect alignments. Delimiters might be a problem

5.2 Limitations

Give a short introduction to the different limitations of our method.

Mention that the scoring matrix could be improved to use a normal distribution for scoring numerical differences.

Mention the problem with our restricted version of multiple sequence alignment that it the longest message could optimally need gap insertion as well.

5.2.1 Textual protocols

Explain why our method is not applicable to textual protocols.

5.2.2 Variable number of fields

Explain the problem with protocols that contain a variable number of fields and why our method does not accommodate for them. (Because we based our definition of protocols on a fixed number of fields?)

5.2.3 Non-aligned data

Explain why we only try to find aligned fields and the problem with complexity if we were to relax this requirement.

5.2.4 Bit precision

Talk a little bit about how some protocols do not use entire bytes as the sizes of their fields and that we will not find exact boundaries for them.

5.3 Future work

Mention the ideas that has come up during the thesis work that we have not had time to investigate further.

5.3.1 Correspondence analysis

Explain how correspondence analysis could give a better result than PCA.

5.3.2 Timestamp identification

Explain how timestamps could potentially be identified from their distinguished byte distribution pattern.

Chapter 6

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