

Automatic Field Extraction of Extended TLV for Binary Protocol Reverse Engineering

Zewen Huang¹, Kui Wu¹, Shengqiang Huang², Yang Zhou², Ronnie Salvador Giagone²

¹Department of Computer Science, University of Victoria, B.C., Canada

²Huawei Technologies Canada Co. Ltd., Burnaby, B.C., Canada

Abstract—Type Length Value (TLV) is one of the main structures commonly used in network protocols. A large number of proprietary protocols, whose specification is unknown to the public, run in the current Internet as well as domain-specific Internet of Things (IoT) applications. It is critical to infer the TLV fields within a packet because this information can help network administrators quickly identify abnormal traffic and potential attacks. Inferring TLV fields belongs to the general task of protocol reverse engineering and is particularly challenging for binary protocols, where the boundaries of TLV fields have many possible positions. Existing methods for reverse engineering binary protocols involve many parameters and only work for protocols strictly following the conventional TLV format. We extend the concept of TLV to accommodate a broader category of structural patterns in various binary protocols, such as TCP, IP, ModBus, and MQTT. We then design algorithms to *automatically* extract the extended-TLV fields from packets. Via a series of experiments over several protocols, we demonstrate that our algorithms can accurately and quickly identify the extended-TLV fields in all the tested protocols. Our approach can thus be deployed as a general method for automatically reverse engineering binary protocol format.

Index Terms—Extended TLV, Binary Protocol Reverse Engineering.

I. INTRODUCTION

Over 40% of traffic on the backbone of the Internet is using private protocols [1], among which a significant portion are binary protocols. This phenomenon becomes more prominent with the recent advance of the Internet of Things (IoT), where many IoT systems are proprietary and use protocols unknown to the public. The lack of knowledge of underlying network protocols poses a big barrier to the accurate detection of abnormal traffic and intrusions. Network protocol reverse engineering (PRE) is one critical technique to address this problem.

The primary purpose of PRE is to automatically deduce the underlying message format of a network protocol without knowing any knowledge about the protocol specification. A network protocol can be classified as text-based, such as HTTP, where packet header includes plaintext and is easy to analyze, and binary-based, such as TCP, where packet header contains only binary bits and it is challenging to identify the boundaries of each field. PRE techniques generally fall into three categories, network-based, program-based, and hybrid methods [2]. The first category analyzes network traffic; the second analyzes the behaviour of software components at the end hosts; the third uses both. We focus on network-based

PRE in this work, because it is the most viable solution for many real-world systems where we cannot access the end hosts for various reasons. For instance, most smart camera manufacturers offer web-based surveillance services where the remote server is located in the cloud and controlled by the service provider.

Among previous works in automatic reverse engineering private protocols, the most famous one is the Project Informatics (PI) project [3]. PI borrowed the idea from bioinformatics and used a multiple sequence alignment (MSA) algorithm to discover the common pattern across the packets in network traffic. Since then, the concept of sequence alignment has played a fundamental role in PRE and has motivated a series of subsequent works. For instance, Cui et al. [4] designed Discoverer, a tool that tokenizes the sequence of packets and uses type-based sequence alignment and clustering algorithms to infer protocol message formats. Tao et al. [5] proposed PRE-bin, which uses the Bayes decision model to determine field boundaries in binary protocols.

All the above methods, however, have drawbacks when applied in PRE binary protocols. The MSA algorithm used in PI compares sequence *byte by byte* whereas fields of a binary protocol may only contain a few bits. Discoverer [4] defines a token as a sequence of consecutive *bytes* that are likely to belong to the same application-level field. Tokenization depends on printable characters, which usually do not appear in binary protocols. Indeed, the authors pointed out that “for binary fields, identifying field boundaries is very hard; so we instead simply declare a single binary byte to be a binary token in its own right.” PRE-Bin [5] tried to solve the problem of identifying field boundaries in binary protocols by using multiple sequence alignment and the Bayes decision model. PRE-BIN, however, introduces many parameters that need a manual tune-up, making PRE-BIN extremely hard to use, particularly when people have no prior knowledge of which parameters work empirically the best.

We tackle the problem of inferring field boundaries in binary protocols from a fresh new angle. So far, all the methods try to generalize for all kinds of packet formats without relying on prior knowledge. It is possible and definitely helpful to learn some prior knowledge by studying the protocol fields that commonly appear in many known protocols. We omit ASCII fields since they can be accurately identified with existing methods [3]–[5] and mainly focus on binary fields. To this end, we start with a broadly-used data structure, type-length-

value (TLV), as prior knowledge and generalize it to *extended TLV*, whose formal definition will be given later, to cover a large corpse of protocol header formats.

A. Challenges

Generalizing the concept of TLV to cover a broader category of protocols and automatically identifying the (extended) TLV fields pose several challenges that cannot be solved with existing PRE techniques.

- Extended concept about L : In the traditional version, L is a length field measuring the size of a block called value (V) field that follows *right after* the L field. Many protocols do not conform to this traditional definition. For instance, in TCP header, the header length has 4 bits in the unit of 4 bytes (e.g., 1010 means the header length is $40(= 10 * 4)$ bytes. To generalize TLV for TCP header analysis, L is still a length field; However, the V field can be any block of fields and may be located anywhere in the packet as long as its relative distance from L is fixed. In addition, we need to allow L to measure the size of the V field plus some offset since we do not know what and where the V field is.
- Binary protocols: Since we are dealing with a binary protocol, any bit could be the starting bit of a TLV field, which may only contain a few bits. Inferring boundaries in the unit of bits is extremely challenging and time-consuming.
- Unit problem: The unit of the L field is uncertain. Assume that the size of V field uses 32 bits and the L field measures the size of the V . The L field will be 1 if L is measured in the unit of 4 bytes (like TCP), and will be 100 if L is measured in the byte unit. Hence, a different unit will lead to a different length of the L field. Unfortunately, for a private binary protocol, we do not know in advance the unit used in the measure.

In this research, we present a solution to tackle all the above challenges. The key idea is based on the following fact: the relative distance between the L and the V fields is usually constant for a given protocol. By identifying this constant across all packets, we could spot the possible locations for extended-TLV fields. Although calculating this constant requires the value of the L field and any bit could be the first bit of the L field in a binary protocol, the length of the L field must be bounded. Due to this reason, we can deduce not only the value of the L field but also the unit it uses. Leveraging this idea, we make the following contributions:

- We formally define the concept of extended TLV, in which the L field could potentially measure anything within the packets. In addition, we define a more general packet structure where the extended TLV can be applied. Not only does it cover the traditional TLV, but it can also extract the L field even when there is a constant offset in the V field.
- We systematically analyze the efficiency of our algorithm, called *Auto-ETLV*, in terms of complexity, converge rate,

and accuracy. The algorithm contains two main operations, Generation and Verification. Generation generates and records all sets of possible values for our interests, and Verification quickly verifies the percentage of packets for which the recorded values hold. The final output will be a set of possible values. We also analyze the converge rate, i.e., how quickly the set of possible values converges to a small number of values.

- We implement and test *Auto-ETLV* over several binary protocols, including IP, TCP, Modbus, and MQTT. Our results show that it is efficient in terms of running time. It can achieve high accuracy by identifying the exact L field for Modbus and returning a number of possible locations for the L field for TCP.

The rest of the paper is organized as follows. We introduce related work in Section II. In Section III, we formally define the concept of extended TLV and the notations we will use throughout the paper. In Section IV, we show some useful observations in the binary protocol, based on which we design algorithms for automatic field identification in Section V. We evaluate our algorithm in Section VI and discuss the potential applications of our work in Section VII. The paper is concluded in Section VIII.

II. RELATED WORK

The current PRE methods mainly fall into two categories, network trace-based approach and execution trace-based approach. The former takes traffic captured on the network as input, whereas the latter assumes the programs running the protocol are accessible and takes the program binary code as input. Depending on how we process the input and output, we can further divide them into active/passive or protocol format/grammar [6]. The active method purposely sends messages to the network to create traffic, whereas the passive method only captures the traffic on the network without creating traffic. On the other hand, the output of PRE could be either a protocol structure/format which specifies the protocol fields or a grammar that includes the rules for message exchanges.

A. Network Trace-Based Approach

The PI [3] project developed idea of Multi-Sequence Alignment (MSA) which composed of Needleman-Wunsch [7] and Smith Waterman [8] algorithms. Packets are classified, and protocol formats are generated by aligning similar packets. Nevertheless, the PI project mainly identifies similar packets with clustering, and it needs users to determine the protocol fields manually. One automatic method that produces protocol format, called Discoverer [4], breaks each packet into a sequence of tokens, then recursively classifies the tokens. In the end, different token types are interpreted as different protocol fields. Another similar method, called ReverX [9], keeps a set of predefined delimiter characters and uses it to break a message into fields. It constructs two automata to model the protocol fields and the protocol grammar. Both ReverX and Discoverer only work on byte-aligned protocols and require

a predefined symbol set. Another automatic method, PRE-Bin [5], handles the non-byte-aligned binary protocols by combining multi-sequence alignment and the Bayes’ decision model. PRE-Bin, however, requires users to configure up to eight parameters manually.

B. Execution Trace-Based Approach

Polyglot [10] produces a protocol format and analyzes the binary code using a shadowing algorithm for PRE that collects all necessary information when the protocol program is running. AutoFormat [11] extracts different protocol fields by examining different execution contexts such as run-time call stack for every message. MACE [12], on the other hand, builds a finite state model that learns from execution loops to infer the protocol grammar.

In summary, the network trace-based approach takes network traffic as input, whereas the execution trace-based approach takes program code as input. Our work belongs to the former but differs from any existing ones. Unlike most network trace-based methods relying on classification, we define a general structure for TLV inference and directly identify the fields based on the most-likely matches in network traffic data.

III. BACKGROUND AND PROBLEM DEFINITION

A. What is TLV?

Type-length-value (TLV) is a broadly-used scheme for encoding information in communication protocols. It’s also referred to as tag-length-value or key-length-value. TLV-encoded packets include the type code, followed by the length of value, and then the value itself. This encoding scheme has been used in a large class of protocols such as SSL/TLS [13], RADIUS [14], and LLDP [15]. A canonical form of the TLV encoding scheme uses *consecutive* blocks as follows:

$$P = |\dots|T|L|V|\dots| \quad (1)$$

where P denotes a packet, T denotes the type code, L denotes the length field, V denotes the value field, and \dots represents the rest parts of the packet.

As network protocols evolve, TLV may appear in various forms. For example, the scheme may appear as LTV [16], which is the same as TLV except that the positions of L and T are swapped. As another example, the CoAP protocol [17] uses a TLV-like encoding scheme, where V is a token field, and L measures the length of the token. But V does not immediately follow the L field, and there are other fields between the L field and the V field. In this example, the L field and the V field are not consecutive blocks. They are separated with a fixed number of bits in between. If we relax the canonical definition of TLV, we can cover a much broader category of protocols into this “TLV” encoding scheme. For example, if we relax the requirement by assuming that the V field is as large as the packet header¹, TCP header can fit into the relaxed

¹Note that we do not know the boundary between TCP header and data in PRE, but the V field can be inferred if the trace includes diverse TCP packets, some of which may not include data, e.g., TCP SYN and FIN packets.

TLV encoding scheme since the L field measures the entire TCP header.

The above observation motivates us to extend the canonical TLV form, based on which we can build a powerful tool for reverse engineering a large group of unknown protocols.

B. Extended TLV (ETLV)

To help better understand our extended TLV scheme, we first introduce several functions.

- Length function $\mathcal{L}(x)$: It determines the number of bits used to represent binary string x , e.g., $\mathcal{L}(101) = 3$.
- Value function $\mathcal{V}(x)$: It determines the value of binary string x in decimal, e.g., $\mathcal{V}(101) = 5$.
- Binary function $\mathcal{B}(d)$: It converts decimal d to binary string, e.g., $\mathcal{B}(5) = 101$.

With the above functions, we formally define the extended TLV (ETLV) as follows.

Definition 1. Extended TLV (ETLV): Assume that P is a packet in the format of $P = |F_1|L|F_2|V|F_3|$, where $\mathcal{L}(F_1)$, $\mathcal{L}(L)$, $\mathcal{L}(F_2)$, and $\mathcal{L}(F_3)$ are all constant. In addition, if P satisfies any of following conditions:

- 1) Case 1: $\mathcal{V}(L) = \mathcal{L}(V)$,
- 2) Case 2: $\mathcal{V}(L) = \mathcal{L}(P)$,
- 3) Case 3: $\mathcal{V}(L) = \mathcal{L}(V) + C$, where C is a constant (offset),

we say that the packet P follows the extended TLV (ETLV) structure.

In the ETLV packet structure, we introduce three blocks F_1, F_2, F_3 , each of which may contain any number (including zero) of bits. The F_1 block means that the L field locates at a fixed location relative to the start of the packet since the L field usually does not appear at the beginning of a packet. The F_2 block separates the L field and the V field so they can stay apart by a fixed distance. The F_3 block allows any extra fixed-size information at the end of the V field, since a protocol may not end with the V field. Note that we do not explicitly define a T field, implying that the T field could be in any of the F_1, F_2, F_3 blocks. In addition, we assume that F_1, F_2, F_3 , and L all have fixed length, including zero. In particular, when the size of L is zero, i.e., $\mathcal{L}(L) = 0$, we have $\mathcal{L}(V) = 0$ since we do not have a L field. In this special case, the packet P would only contain F_1, F_2, F_3 , suggesting that P has a fixed size.

Next, we use examples to explain different cases in the ETLV structure.

- Examples of Case 1: This case is similar to the canonical TLV, where the L field tells the length of the V field. All protocols following the canonical TLV are covered by this case. For example, in ModBus and MQTT protocol [18] the value of the length field measures the size of remaining fields.
- Examples of Case 2: This case allows the L field to measure the size of the entire packet. One example is the IP protocol where the length field tells the size of the whole packet.

- Examples of Case 3: This case covers special situations where the L field may not measure the size of the V field exactly; instead, there is a fixed gap. For example, in CoAP, there are two extra fields between the token length field and the token field. On the other hand, if the V field contains some data, and the L field specifies the size of a buffer that the receiver should allocate, $\mathcal{L}(V)$ could be larger than the actual size of the V field. Some proprietary protocols fall in this category [19].

Remark 1. We assume that a given protocol can only follow one of the above cases, but we do not know exactly which case the protocol follows. This assumption is reasonable since a protocol usually grants the unique meaning of an L field.

Remark 2. The definition of ETLV uses the \mathcal{V} function, which assumes the unit in bits. Yet, we DO NOT know in advance the unit of the L field used in an unknown protocol, so we cannot tell its value $\mathcal{V}(L)$ in terms of bits. For instance, the L field in the TCP header is measured in the unit of 4 bytes. Therefore, the three cases described above could all fail if we cannot automatically infer the right unit used in the protocol. We solve this problem in our later algorithms.

With ETLV defined and explained, our goal in the rest of the paper is to develop a tool, called, *Auto-ETLV*, that automatically extracts the L field and the V field in an unknown binary protocol following the ETLV packet structure (Sections IV and V). We also evaluate the accuracy of *Auto-ETLV* (Section VI).

IV. PROBLEM ANALYSIS

To automatically extract the information about the L and V fields from an unknown binary protocol, we need to tackle the following challenges. First, the boundaries of each field are unknown and may appear at any bits in a packet. Second, different protocols may use different units when calculating the value of the L field. Third, while we assume that a given protocol can only use one of the three cases in ETLV, we do not know whether or not the protocol actually follows, and even it does, we do not know which exact case of ETLV it uses.

To address the above challenges, we first analyze the properties of the \mathcal{V} and \mathcal{L} functions. Based on the analysis, we then solve the unit problem when calculating the value of the L field. Finally, we develop a method to find out the most likely ETLV case that the protocol follows.

A. The Property of $\mathcal{V}(\hat{L}), \mathcal{L}(\hat{L})$

Usually not all bits in the length field are used to represent the size of the value. For example, to represent a value 7, the L field of 5-bits would be 00111. In other words, only the last three bits are used for calculation. We call the **effective length field** \hat{L} as the binary string of the L field excluding preceding 0s. Note that $\mathcal{V}(L) = \mathcal{V}(\hat{L})$.

Lemma 1. If a protocol uses ETLV and the effective length field \hat{L} is in unit of bits, then any packet P satisfies the following constraint:

$$\mathcal{L}(\hat{L}) < \mathcal{L}(\mathcal{B}(\mathcal{L}(P))) + 1, \quad (2)$$

where \mathcal{L}, \mathcal{V} , and \mathcal{B} are the functions defined in Section III-B.

Proof. Since the protocol follows ETLV, we only need to verify Equation (2) in the three cases defined by ETLV.

- 1) Case 1: $\mathcal{V}(L) = \mathcal{L}(V) < 2\mathcal{L}(P)$.
- 2) Case 2: $\mathcal{V}(L) = \mathcal{L}(P) < 2\mathcal{L}(P)$.
- 3) Case 3: $\mathcal{V}(L) = \mathcal{L}(V) + C < \mathcal{L}(P) + \mathcal{L}(P) = 2\mathcal{L}(P)$.
Note that C is a constant offset and should not be larger than $\mathcal{L}(P)$.

Therefore, we have $\mathcal{V}(L) < 2\mathcal{L}(P)$ in all cases. Thus, the number of bits that are needed to represent $\mathcal{V}(L) = \mathcal{V}(\hat{L})$ is at most the number of bits used to represent $2\mathcal{L}(P)$, or equivalently, $\mathcal{L}(\hat{L}) < \mathcal{L}(\mathcal{B}(\mathcal{L}(P))) + 1$. \square

Note that even if L is not measured in bits, Lemma 1 still holds since $\mathcal{V}(L)$ can only become smaller if measured in other units.

B. The Unit Problem

Based on Lemma 1, we know the upper bound of $\mathcal{L}(\hat{L})$. But we still do not know how to determine the actual unit of the L field. Without knowing its unit, we cannot determine $\mathcal{V}(L)$ and thus cannot tell if a packet falls into any of the three cases of ETLV.

We use two examples to illustrate the unit problem. If a field F uses 64 bits and the corresponding L field tells the length of this F field, then $L = 1000000$ if the unit of L is bits, $L = 100000$ if the unit is 2 bits, $L = 10000$ if the unit is 4 bits, $L = 1000$ if the unit is in bytes, and so on. If the field F uses 40 bits and the corresponding L field tells the length of this F field, then $L = 101000$ in bits, $L = 10100$ in 2 bits, $L = 1010$ in 4 bits, $L = 101$ in bytes, and so on. From the above examples, we can see that no matter what unit a binary string uses, only the number of trailing zero's matters. Everything before the trailing zeros is unchanged. We call the bits before the trailing zeros **core bits**. The core bits in the first example are 1, and the core bits in the second example are 101.

Therefore, given a L field binary string, we can shift its **core bits** to the left to get the length values in different units. The good news is that $\mathcal{L}(\hat{L})$ is bounded based on Lemma 1. Thus the number of possible shifts/units that we need to check is also bounded. If any shift matches one of three cases across all packets, then we know this shift value implies the most-likely unit used by the L field, which is in 2^{shift} bits.

C. The Property of $\mathcal{V}(L), \mathcal{L}(\hat{L})$ in Different Units

We refer the *shift* mentioned above as *unit size*, e.g., unit size = 0 means that the \hat{L} field is measured in $2^0 = 1$ bit. Lemma 1 tells the bound of the length of the \hat{L} field in the unit of bits. Next, we show that a tighter bound if a different unit is used.

Lemma 2. *If a protocol uses ETLV, then the effective length field \hat{L} of packet P , satisfies the following constraint:*

$$\mathcal{L}(\hat{L}) < \mathcal{L}(\mathcal{B}(\mathcal{L}(P))) - \text{unit size} + 1, \quad (3)$$

no matter what unit \hat{L} uses.

Proof. No matter what the unit \hat{L} uses, if we shift \hat{L} to left with *unit size*, $\mathcal{V}(\hat{L} \ll \text{unit size})$ will return the value of \hat{L} in bits. In other words, $\mathcal{L}(\hat{L}) + \text{unit size} = \mathcal{L}(\hat{L} \ll \text{unit size})$ since both measures the same field in bits. For instance, if \hat{L} uses byte as the unit and $\hat{L} = 100$ (i.e., 4 bytes), then $\mathcal{V}(\hat{L} \ll \text{unit size}) = \mathcal{V}(100000) = 32$ (i.e., 32 bits). Since Lemma 1 is true when the unit is in bits, we have

$$\mathcal{L}(\hat{L}) + \text{unit size} = \mathcal{L}(\hat{L} \ll \text{unit size}) < \mathcal{L}(\mathcal{B}(\mathcal{L}(P))) + 1. \quad \square$$

D. Constant Difference

From the previous analysis, given an L field in a binary string, we may have a small set of shifted values, and one of them will be the correct version of $\mathcal{V}(L)$ in terms of bits. We have the following lemma that can help us make the right decision.

Lemma 3. *Assume that a protocol uses ETLV and its packet $P = |F_1|L|F_2|V|F_3|$. Denote the blocks after the L field as $R = |F_2|V|F_3|$. Then, $\mathcal{L}(R) - \mathcal{V}(L)$ is a constant.*

Proof. We only need to verify the three cases defined in ETLV.

1) Case 1: Since $\mathcal{V}(L) = \mathcal{L}(V)$, we have

$$\begin{aligned} \mathcal{L}(R) - \mathcal{V}(L) &= \mathcal{L}(F_2VF_3) - \mathcal{L}(V) \\ &= \mathcal{L}(F_2) + \mathcal{L}(V) + \mathcal{L}(F_3) - \mathcal{L}(V) \\ &= \mathcal{L}(F_2) + \mathcal{L}(F_3) \end{aligned}$$

2) Case 2: Since $\mathcal{V}(L) = \mathcal{L}(P)$, we have

$$\begin{aligned} \mathcal{L}(R) - \mathcal{V}(L) &= \mathcal{L}(R) - \mathcal{L}(P) \\ &= (\mathcal{L}(P) - \mathcal{L}(F_1L)) - \mathcal{L}(P) \\ &= -\mathcal{L}(F_1) - \mathcal{L}(L) \end{aligned}$$

3) Case 3: Since $\mathcal{V}(L) = \mathcal{L}(V) + C$ where C is a constant, we have

$$\begin{aligned} \mathcal{L}(R) - \mathcal{V}(L) &= \mathcal{L}(R) - (\mathcal{L}(V) + C) \\ &= \mathcal{L}(F_2) + \mathcal{L}(V) + \mathcal{L}(F_3) - (\mathcal{L}(V) + C) \\ &= \mathcal{L}(F_2) + \mathcal{L}(F_3) + C \end{aligned}$$

We can see that no matter which case a packet follows, $\mathcal{L}(R) - \mathcal{V}(L)$ always remains constant. \square

Utilizing the above analytical results, we next develop the *Auto-ETLV* tool that automatically extracts the L field from a binary protocol if it follows the ETLV structure.

A. A High-Level Overview

For an unknown protocol, the L field may appear anywhere in a packet, and as such, we have to loop through each bit of the entire packet. Based on previous analysis, for each bit, we only look at $\mathcal{L}(\mathcal{B}(\mathcal{L}(P))) + 1$ consecutive bits (called *check string* and denoted by cs), since $\mathcal{L}(L)$ is bounded by this number. Then we generate each suffix from cs . For each suffix, we can apply the left shift operation to get different possible values for different units. Then we use the shifted string to calculate $\mathcal{V}(\text{shifted string}) - \mathcal{L}(\text{remaining bits})$, and we call this value the *key difference*. If all packets have the same *key difference* at a certain bit position ($bpos$) and certain shift number. Then we can infer that this $bpos$ is a possible location for the length field, with the unit in 2^{shift} bits.

While the above idea seems straightforward, it requires us to generate many *key differences* for each location. Denote $b = \mathcal{L}(P)$. We then need to loop through b bits. For each bit, we look $\log(b)$ bits *check string* and generate $\log(b)$ suffixes. For each suffix, we have $\log(b)$ shifts. Thus we have $O(b \log(b)^2)$ possible *key differences* for every packet. This is a huge number! Nevertheless, we will show that as the number of packets fed to the algorithm increases, this number will drop dramatically with a proper filtering mechanism to exclude the impossible bit positions.

Next, we introduce the algorithm details for extracting the length field from ETLV. The algorithm is mainly composed of two steps: the Generation (G) step and the Verification (V) step, and only one of them is executed for each packet. The G step is executed with small probability since it takes most of time to generate all *key differences* at every bit position for the L field. The V step is executed with high probability since it only verifies what we have generated from the G step and filters out impossible bit positions. The above generation-verification method can reduce the running time and the number of possible bit positions for the L field dramatically.

B. Global Dictionary

The core data structure we use is the global dictionary (GD). Each entry in the GD has a key field and a value field. The key field consists of $(bpos, shift)$, and the value field consists of $(index, Kcounter, a \text{ list of tuples})$, where each *tuple* is represented by $(key \text{ difference}, l, r, Dcounter)$.

Recall that we loop through all bits in a packet, and for each bit position $bpos$, we have a *check string*, with which all suffixes are generated. For each suffix, we apply the left shift operation to generate the values of the binary string of this suffix in different units. Then we use each value of the shifted suffix to calculate the *key difference*. Note that for each suffix, they can be shifted different times, thus for each $(bpos, shift)$ combination, we will have a set of *key differences*. We need to store all the information for every $(bpos, shift)$ combination. Thus we use $(bpos, shift)$ as the key of the GD.

The value field includes a list of *tuples*, each denoted by $(key \text{ difference}, l, r, Dcounter)$. Note that *key difference* is

Algorithm 1: Main Structure

```
input : a set of packets
output: the possible locations for the  $L$  field
begin
  GD=;
  for  $index, p$  in  $enumerate(Packets)$  do
    if  $uniform(0, 1) < \frac{1}{index+1}$  then
       $\lfloor$  Generation( $p, index$ );
    else
       $\lfloor$  Verification( $p, index$ );
   $\lfloor$  return GD.maxDCounter();
```

calculated by $\mathcal{V}(\text{shifted string}) - \mathcal{L}(\text{remaining bits})$, and l and r together denote the inferred interval (i.e., the left bit position and the right bit position, respectively) for the left boundary of the L field. We also include $index$ to record the timestamp when this GD entry is created and $Kcounter$ to record the number of times that the *list of tuples* has been updated. $Dcounter$ is another counter recording the number of times that the same *key difference* value has been encountered.

C. Main Structure of Auto-ETLV

Algorithm 1 shows the main structure of our solution. It first creates a global dictionary to store all information for possible length field locations generated from the G step. In the main loop, the algorithm chooses one of the two steps for each packet with probability. As the number of packets increases, the probability that we choose the G step becomes smaller. It is worth noting two facts. First, the V step only verifies all possible locations generated from the G step, so the V step will not take any effect until we run the G step. Second, we use a dynamically adjusted probability for executing the G step, i.e., $\frac{1}{index+1}$ in Algorithm 1. While this method leads to good experimental results, users may use a different probability value. No matter which probability value to use, we need to ensure that the G step is executed at least several times because otherwise, there is nothing to verify in the V step.

In the end, the $Dcounter$ reflects the likelihood that the represented bit location is the correct location of the L field. After processing all packets, we report the location(s) that have the highest $Dcounter$ value(s).

D. Generation Step

In the Generation step (Algorithm 2), we loop through all bits in a packet. For each bit position ($bpos$), we calculate all *key differences* for different possible shifts. Since the L field is bounded, the check string (cs) for generating possible differences is also bounded. If we assume that the current $bpos$ is the bit right after the L field, then the possible L fields are all suffixes of cs. Furthermore, we need to infer the unit of the length field. The good news is that the L field is bounded; hence we can determine all possible values for all possible units by shifting the suffix to the left, bit by bit, for the length of cs times. After this, we have all possible values for possible

Algorithm 2: Generation

```
input :  $p$ : a packet
        index:  $i$ -th packet
begin
   $cl = \mathcal{L}(\mathcal{B}(\mathcal{L}(p))) + 1$ ;
  for  $0 \leq bpos \leq \mathcal{L}(p)$  do
     $cs = p[\max(0, bpos-cl), bpos]$ ;
     $r = \mathcal{L}(P) - bpos$ 
    suffix_set = a set of suffixes of cs
    diff_dic = {}
    for  $suffix$  in  $suffix\_set$  do
      for  $0 \leq shift \leq cl - \mathcal{L}(suffix)$  do
        key = ( $bpos, shift$ )
        item = ( $\#$ Of preceding zeros of suffix,
               $\mathcal{L}(suffix)$ ,
               $\mathcal{V}((suffix \ll shift) - r)$ ), 1)
         $\lfloor$  diff_dic[key].append(item)
    for  $key, val$  in  $diff\_dic$  do
       $\lfloor$  GD.update(index, key, val, append = True )
```

units at this $bpos$. Accordingly, we can calculate and update the difference to the global dictionary.

Note that in the G step, we need to consider a list of (*key difference, l, r, Dcounter*). The fields l, r together denote the inferred range of possible length for the L field. Intuitively, since each suffix could be a potential L field, the length of the true L field is at least as large as the length of the suffix. On the other hand, not all bits in the L field are usually used in a packet (e.g., the first several bits of the L field are 0's). Thus the length of the true L field is at most as large as the length of the suffix plus the preceding zeros. The *key difference* is calculated by $\mathcal{V}(\text{shifted string}) - \mathcal{L}(\text{remaining bits})$, and $Dcounter$ counts how many times we have encountered the same *key difference* value. The timestamp records when this key was added to the global dictionary for the first time.

We also allow $bpos$ to be as high as the length of the packet ($\mathcal{L}(p)$). This is because $bpos$ is assumed to be the bit right after the L field, and we need to handle the (rare) case that a packet ends with the L field, i.e., no bit after the L field. Allowing $bpos$ to be $\mathcal{L}(p)$ will not cause any problem, since we only process bits right before the $bpos$.

E. GD update

Algorithm 3 will be called in both the G and V steps. The only difference is that this function could increase the size of the list of *tuples* under the given key in the G step, but not in the V step. To update the GD, if the key does not exist in GD, we set its first part of dictionary value as ($index, 1$), where $index$ works as the timestamp indicating when this key is created and 1 is the current value of $Kcounter$. Note that we use packet index as the timestamp because they mean the same thing for packet trace analysis. If the key already exists in the GD, we first increase $KCounter$ by 1. Then we check if the

Algorithm 3: GD.update

```
input : index:  $i$ -th packet
         key: (bpos, shift)
         val: a list of key difference
         append: bool variable

begin
  if key in GD then
    for  $v$  in val do
      for  $gv$  in GD[key][1] do
        if  $v.diff == gv.diff$  then
           $gv.l = \min(gv.l, v.l)$ 
           $gv.r = \max(gv.r, v.r)$ 
           $gv.Dcounter++$ 
        else if  $append == True$  then
          GD[key][1].append( $v$ )
      if  $gv$  ever updated then
        GD[key][0].Kcounter++
  else
    GD[key] = ((index, 1), val)
```

list of tuples already includes the same *key difference*. If yes, then we increase the corresponding *Dcounter* by 1 and update l and r . Otherwise, we add a new *tuple* ($(key\ difference, l, r, 1)$) to the list of *tuples*, where *Dcounter* is set to 1.

To update l and r , imagine that we have two L fields from two packets. Let s_1 and s_2 be such two binary strings representing the two L fields. Then the length of the L field will be at least $\max(\mathcal{L}(s_1), \mathcal{L}(s_2))$, and at most $\min(\mathcal{L}(\text{preceding zeros} + s_1), \mathcal{L}(\text{preceding zeros} + s_2))$.

F. Verification Step

The verification step (Algorithm 4) is almost the same as the G step, except that we only verify what we have generated from the G step. Thus we only loop through the global dictionary. Since the key contains $bpos$ and $shift$, we can generate all values for the specific unit calculated from the $shift$ value. Then we can use the values to calculate the differences.

We use an additional condition to clear up GD at the end of the verification step. The *Kcounter* tells the number of times we have encountered this key in the global dictionary. The *index* records the first time that this key was added to GD. The last condition check in Algorithm 4 determines whether the number of encountered times is less than 90% of times since it appeared the first time. The value of 90% is empirical. The idea is that if this key is not encountered often, then this key should be dropped from GD. Note that if the L field matches the size of the V field in all packets, then the number of times that the key encountered for the correct L field passes the verification should be 100%. If any packet has the L field mismatching the V field, then the number of times the key encountered for the correct L field will not be 100%. Note that

Algorithm 4: Verification

```
input : p: a packet
         index:  $i$ -th packet

begin
   $cl = \mathcal{L}(\mathcal{B}(\mathcal{L}(p)))+1$ ;
  for  $bpos, shift$  in GD.key do
     $cs = p[\max(0, bpos-cl), bpos]$ ;
     $r = \mathcal{L}(P)-bpos$ 
    suffix_set = a set of suffixes of  $cs$ 
    diff_dic = {}
    for suffix in suffix_set do
      if  $shift \leq cl - \mathcal{L}(\text{suffix})$  then
        key = (bpos, shift)
        item = (#Of preceding zeros of suffix,
               $\mathcal{L}(\text{suffix})$ ,
               $\mathcal{V}((\text{suffix} \ll shift) - r)$ , 1)
        diff_dic[key].append(item)
    for key, val in diff_dic do
      GD.update(index, key, val, append = False)
      timestamp = GD[key].Index
      hit = GD[key].Kcounter
      if  $(index - timestamp)*0.9 > hit$  then
        GD.remove(key)
```

this cleanup step is to speed up future processing by reducing the size of GD.

G. Time Complexity

We analyze the time complexity of processing each packet. The required storage space is essentially the size of the global dictionary. The space cost is negligible, considering that the number of possible combinations for the key field is not large for any given protocol.

Let b be the size of the largest packet, d be the number of entries (keys) in the global dictionary, and k be the largest number of tuples in the value fields of the global dictionary.

1) *Complexity of the G Step*: In the G step, we need to loop through all the bits, and for each bit, we need to generate all possible suffixes of cs . Since cs is bounded by $\log(b) + 1$, we have $\log(b) + 1$ suffixes and it takes $O(\log(b)^2)$ time to process them. For each suffix we can shift $\log(b) + 1 - \mathcal{L}(\text{suffix}) = O(\log(b))$ times to get different values in different units, that are used to calculate the *key differences*. So it will take $O(\log(b)^2)$ time.

For updating the l value, it requires the knowledge about preceding zeros at current $bpos$. However, this value can be pre-computed by simply looping through all bits once, which takes $O(b)$ time.

For the *GD.update* function, we need to compare the list of *key differences* generated from the current packet and the list of *key differences* in the GD. Since the maximum size of the former list is $O(\log(b)^2)$, and the maximum size of the latter list is $O(k)$, the comparison takes $O(k \log(b)^2)$ time.

In total, the G step takes $O(bk \log(b)^2)$ time.

2) *Complexity of the V Step*: In the V step, we need to loop through all entries in the global dictionary. Since d is the number of entries in the global dictionary, we can follow the similar analysis as in the G step to conclude that the V step takes $O(dk \log(b)^2)$ time.

Note that the V step will be executed more frequently than the G step. In addition, since it can quickly reduce the size of GD due to the filtering mechanism, the value of d is expected to be small.

VI. EVALUATION RESULTS

We test *Auto-ETLV* on several well-known protocols: ModBus, IP, TCP and MQTT. We pretend that we know nothing about these protocols. For each protocol, we collect 2000 packets, which are used as the input to *Auto-ETLV*.

A. Accuracy

TABLE I
LOCATION OF THE L FIELD

Protocol	Proposed Locations (unit)	True Location (unit)
TCP	[(96, 96), 99] (32 bits)	[96, 99] (32 bits)
	[(96, 96), 100] (16 bits)	
	[(96, 96), 101] (8 bits)	
	[(96, 96), 102] (4 bits)	
	[(96, 96), 103] (2 bits)	
	[(96, 96), 104] (1 bit)	
IP	[(21,14),31] (8 bits)	[16,31] (8 bits)
ModBus	[(16, 40), 47] (8 bits)	[32, 47] (8 bits)
MQTT	[(4, 9), 15] (8 bits)	[8, 15] (8 bits)

The L field is represented by [the range of the left boundary, the location of the right boundary].

Table I shows the locations for the L field of a protocol proposed by *Auto-ETLV* and the ground truth. The proposed location is labeled by $[(x_1, x_2), x_3](x_4)$, where (x_1, x_2) denotes the left boundary of the L field (i.e., the left boundary should fall in the range from x_1 -th bit to x_2 -th bit), x_3 denotes the right boundary of the L field, and x_4 is the unit of the L field. The ground truth is denoted by $[y_1, y_2](y_3)$, where y_1 and y_2 are the left boundary and the right boundary of the L field, respectively, and y_3 is the unit of the L field.

From the table, we see that *Auto-ETLV* can produce the correct locations for correct boundaries and units for IP, ModBus, and MQTT. It also produces a small interval containing the correct left boundary of the L field. Note that identifying the exact left boundary of the L field is difficult if no packet in the input data uses the maximum length. In this case, the prefix of the L field will be filled with zeros, and we do not have enough information to figure out the exact location of the left boundary.

It is interesting that *Auto-ETLV* returns six possible locations for the right boundary of the L field in TCP. This is not an error but instead discloses the complex TCP header structure. This result is because TCP has six reserved bits right after the L field, which are set to zeros in all the TCP packets of our input data. In this case, any of 6 bits can be reasonably

interpreted as a valid right boundary of the L field, where different right boundaries correspond to different units, i.e., 1 corresponds to byte, 10 corresponds to 4 bits, 100 corresponds to 2 bits, and 1000 corresponds to bits. In addition, the interval of the left boundary (96, 96) indicates the exact location of the left boundary of the L field.

Note that the TCP header certainly follows ETLV structure since the length field measures the entire header, but TCP does not. There is an extra variable-sized data block that comes after the TCP header. Since it is not a fixed-length block, the key difference will not be computed correctly even if the current $bpos$ points to the correct length field location. However, there is a portion (roughly 30%) of TCP packets that do not come with data (data block with size 0) in the input packets. Thus the $Dcounter$ will be correctly updated with 30% of times. This is high enough to be recognized as a meaningful output as long as we do not filter out this information (controlled by the filtering rate discussed later). From the TCP example, we can see that even if only a portion of packets (e.g., only TCP packets without data block) follow the ETLV structure, our method can still correctly identify the L field.

Auto-ETLV has two parameters to set: the first one is the probability of calling the G step, and the second one is the proportion of the total number of packets that determines when we should drop a key from the global dictionary. The default value for the first is set dynamically to $\frac{1}{1+packet\ index}$, and the default value for the second is set to 0.9. Before we discuss the impacts of these two parameters, we show the dynamic changes of the global dictionary and the maximum $Dcounter$ in the global dictionary.

B. Dynamics of GD

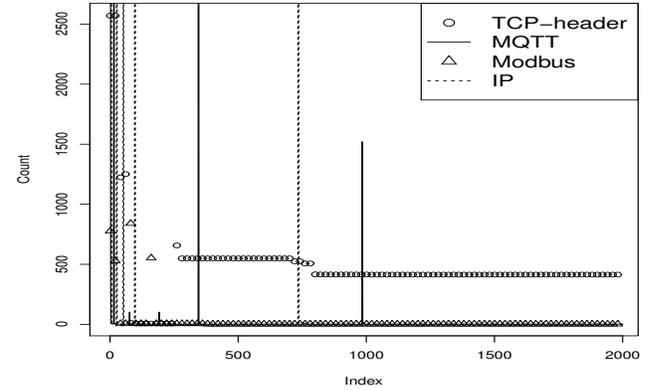


Fig. 1. Size of global dictionary as we process more packets

Recall that we have a filtering mechanism that reduces the size of the GD in the V step. Figure 1 shows the changes in the size of the GD vs. the total number of parsed packets. Every spike in the GD size indicates execution of the G step, since the G step is the only step that generates *key difference*

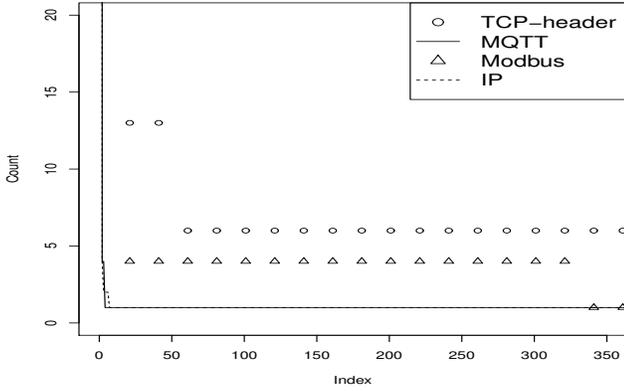


Fig. 2. Maximum Dcounter after processing each packet

information at different $bpos$. We can also see that the GD size drops shortly after every G step.

Although we use 2000 packets for each protocol, it may be unnecessary to use so many packets to get a stable answer. In fact, some protocols only take less than 60 packets to get the final result. Figure 2 shows the size of *key difference* information with the highest Dcounter value for each protocol as the number of parsed packets increases. For MQTT, *Auto-ETLV* only takes 4 packets to reduce 11212 possible *key difference* to the only possible solution, and after that this answer remains unchanged. Of course, the threshold for the number of required packets depends on the diversity of packets, e.g., different packet sizes. Our experimental results suggest that, in general, tens or hundreds of packets should be enough to obtain a stable answer for most protocols.

C. The Impact of Parameters

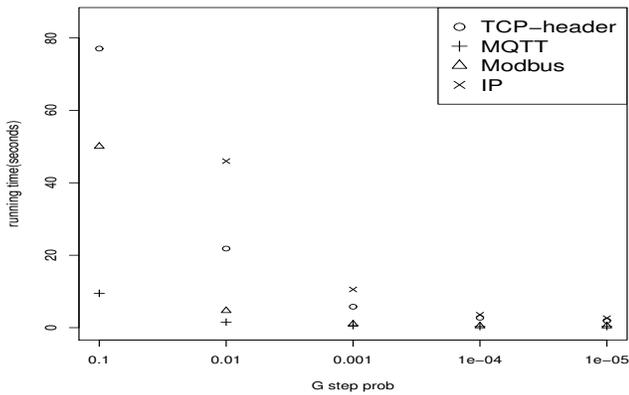


Fig. 3. Running time vs. the probability for executing the G step, the filtering rate = 0.9.

Auto-ETLV uses two parameters: the probability for executing the G step and the filtering rate. The first parameter determines the probability that the G step is executed for a

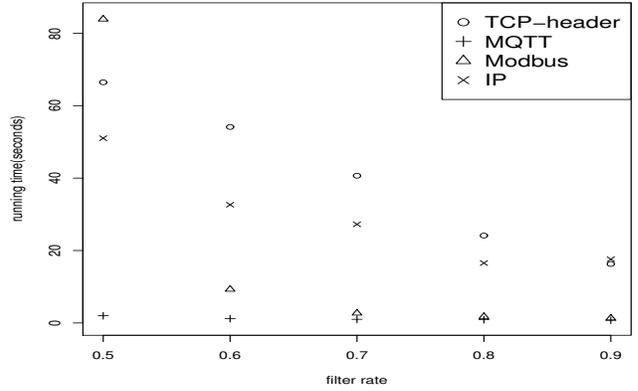


Fig. 4. Running time vs. the filtering rate, the probability for executing G step = $\frac{1}{1+index}$.

packet. The second parameter specifies a proportion value, and if a dictionary key is updated less frequently than this value since this key was created, we drop the key from the GD. To assess the impact of these two parameters on the average running time, we use 1000 packets for each protocol and set the filtering rate to 0.9 when we test the running time of *Auto-ETLV* with different G-step probabilities. In addition, we hold the G-step probability to $\frac{1}{1+packet\ index}$ when we test the running time for different filtering rates. To get the average running time, we run 10 experiments for each parameter configuration on a laptop computer (MacOS, 2.2GHz Dual-Core Intel Core i7, 8GB 1600MHz DDR3). The results are shown in Fig. 3 and Fig. 4. From the figures, we can see that *Auto-ETLV* runs very fast and returns a converged result in less than 2 minutes.

Note that the above two parameters have no impact on the correctness of the algorithm. They only serve as tuning knobs to speed up the algorithm. A natural question is: why should we need to use the G-step probability instead of just running the G step once? We made this decision for precaution. The inputs to the algorithm may contain some corrupted packets where the L field in the packets does not correctly measure the size of its value field. We call these packets *unexpected packets*. If we happen to run the G step on an unexpected packet, then the global dictionary will not contain the correct *key difference* information, and all verification will fail to update the correct *key difference*. To prevent this potential problem, we need to run the G step more than once, and we adopt a probability value that changes depending on the proportion of unexpected packets in the inputs. If the value is set to 1, there is no optimization (w.r.t. running speed), and every packet will use the G step. The result will be accurate but with a much longer running time. On the other hand, if the value is set to 0, then the G step will be executed only once, and the rest packets are processed with the V step. *Auto-ETLV* will run very fast, but the result might be incorrect².

²When the packet processed by the G step is corrupted, *Auto-ETLV* will fail to return right answer.

Regarding the filtering rate, we can set this value to any value rather than 1. When the filtering rate is set to 1, we must drop any key if it is not updated in any packet. This might slow down the algorithm in case we drop a key that needs to be added back to the GD again at a later time. Like the G-step probability, the filtering rate depends on the proportion of unexpected packets in the inputs. When this value is set to 0, the size of GD will never be reduced, and in this case, the V step and the G step will have the same running time.

In summary, if the user can provide a good input where the proportion of unexpected packets is very small, they can choose a large filtering rate and a small G-step probability to greatly speed up the running time.

VII. DISCUSSION

This section explains why ETLV is important and how it can be used in practice. We also discuss the limitation of our work.

As we have illustrated in Section III, ETLV is a pretty general data format that covers the structure of many protocols. The concept of ETLV does not require the TLV fields to be consecutive in a data packet and assumes the default unit of length calculation may vary. Such generality is critical for protocol reverse engineering because we do not know the actual locations of the TLV fields or the unit that the protocol uses to calculate the length. With the identified LV fields, we can leverage this information for many practical applications.

One possible application is for security tests of unknown protocols. Network protocol fuzzing [20] is a technique that sends random data to the protocol port and checks whether or not the protocol responds. With some fields identified with our method, we may effectively generate packets that are likely to follow the legitimate format of the protocol, resulting in more targeted testing of the protocol. Another possible application is to enhance existing clustering-based PRE tools. PRE tools such as those in PI [3] cluster packets based on their similarity. With the help of our solution, the identified fields can be used as a significant feature to validate whether or not the clustering results are reasonable.

While the concept of ETLV and our algorithms are important and useful, they are only one of the building blocks in the arsenal for PRE. Our method alone cannot fulfill the complex task of PRE because real-world protocols usually involve much rich information that ETLV may not model. In many cases, PRE requires expert domain knowledge and human intervention. Our future work will integrate the method presented in this paper into some open-source PRE tools.

VIII. CONCLUSION

In this paper, we extended the concept of TLV to cover the packet structure of a large category of protocols. Based on the extended TLV (ETLV), we proposed an algorithm called *Auto-ETLV* to extract the length field from unknown binary protocols. *Auto-ETLV* not only identifies the left and right boundaries of the length field but also infers the unit that the length field uses. We test *Auto-ETLV* on several

binary protocols, such as TCP, MQTT, and ModBus. The experimental results demonstrate that *Auto-ETLV* can quickly identify the length field for all the protocols under test. The returned answers are correct in the sense that the answers not only include the ground truth but also cover all possible protocol specifications consistent with the parsed packets.

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