A Comparative Analysis of Machine Learning Algorithms for Website Traffic Classification from Network Packets

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ABSTRACT

Accurately Identifying web traffic destination and origins is crucial for the efficiency of a network. This project explores the potential of machine learning in reference to web traffic classification based on the analysis of network packets. We monitored and analyzed web traffic data from ChatGPT, Blackboard, and Linkedin, with the objective of building models which will be able to predict the web traffic origin or destination of a specific packet. The collection of data was performed using Wireshark, then the data was reformatted to eliminate bias and get more accurate results. Using the collected data we then trained four models which had varying levels of accuracy, Logistic Regression (56%), K-Nearest Neighbors (77%), Random Forest (78%), and finally a neural network (85%). This project shows the importance of machine learning within the field of network traffic analysis as automaton is much more efficient and precise compared to manually examining web traffic, especially in a scale as large as the internet. One example of our project's significance is that this can be crucial data analysis for network administrators and security professionals, whom would examine the network for malicious traffic.

KEYWORDS

Network Traffic, Machine Learning, Web Traffic Classification, Network Packets, Data Analysis, Neural Networks, Random Forest, K-Nearest Neighbors, Logistic Regression

1 INTRODUCTION

The ability to monitor and analyze network traffic is crucial for the efficiency and security of a network [4]. It helps to manage the overall network performance, detect and prevent malicious activities, and ensure the network is operating as intended. The present day internet is composed of a vast variety of diverse web traffic, of which requires a more sophisticated approach to analyze and classify network traffic[2]. This project investigates a variety of machine learning algorithms to see which most accurately and effectively classify web traffic based on data from a set of captured network packets.

Traffic classification is very significant in practice, if done accurately and effectively [5]. For reasons mentioned previously, the observed capabilities of this and other projects can be crucial for network administrators and security professionals, whom would examine the network for malicious traffic.

For this project, many sets of packet data were collected from three popular websites: ChatGPT, Blackboard, and Linkedin. These sets were then merged into a single dataset, which was then split into training and testing sets. Four models were then trained and tested on this data: Logistic Regression, K-Nearest Neighbors, Random Forest, and Neural Network [3].

A thorough evaluation of the models was conducted through testing and validation sets, hyperparameter optimization employing cross-validation and grid search, and computing accuracy and Macro F1 scores. The results showed that the Neural Network model achieved the highest accuracy of 85% and Macro F1 score of 84.

These scores display the model's ability to accurately classify the selected websites based on captured network packet data.

Overall, the significance of this project is seen in its use of Machine Learning and the extensions of these applications into the discipline of network traffic analysis. This practice has potential to be utilized in real-world applications, and many sources prove it already is [2].

2 PROPOSED METHOD

This section presents the methodology for how this project goes about capturing network packet data, analyzing the data, and classifying it based on its website of origin.

2.1 Capturing Data

2.1.1 Network analysis tool. For this project, the tool chosen and used was Wireshark [1], a widely used network protocol analyzer. Wireshark is capable of capturing and analyzing network packets, and is able to provide a detailed view of the packets exchanged between the client and server. Wireshark offers a broad variety of features that allow for users to capture, decode, and analyze network packets. It shows



Figure 1: Wireshark Data Collection Process

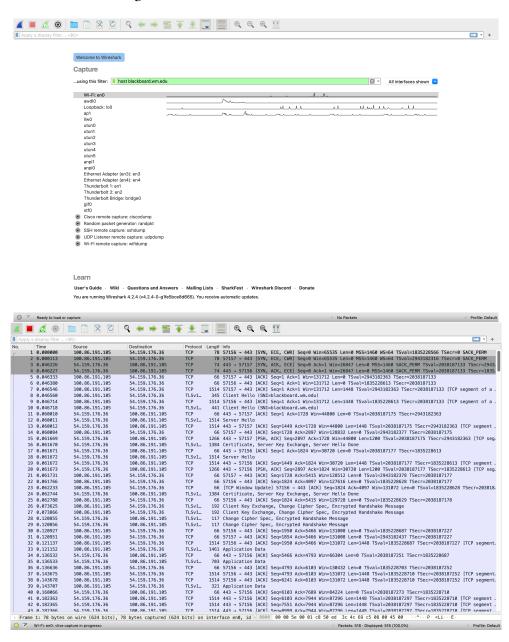


Figure 2: Wireshark interface and example of captured data

the packet data in a human-readable format, and provides a detailed view of the packet's contents. This data contains information on time taken for packet to send and arrive, protocols used, size of each packet, and other fields such as more information on the packet and the source IP address. This tool also supports use of different forms of link communication and can be specified to capture packets from a specific network interface (ie. Ethernet, Wifi, etc.). The many applications and ease of use of Wireshark make it a great tool for this project.

The process for this project (shown in figure 1) involved setting a hostname filter for each desired website, capturing packets from the host, and then exporting the data to a CSV file

2.1.2 Data Collection. To commence the collection of our data, we open Wireshark and select a network connection. In this project we established a connection the College of William and Mary's network interface (en0). In order to isolate the web traffic in order to specifically collect packets that we are analyzing, the capture filter functionality is utilized. The website of interest is then visited in the web browser which causes Wireshark to begin collecting the packets being exchanged. After collecting sufficient packets, the data is then converted into a Comma Separated Values (CSV) file. This process of data collection is performed for Blackboard, LinkedIn, and ChatGPT resulting in the creation of three distinct files.

2.2 Data Preparation and Processing

In order to ensure the integrity and reliability of our models, the collected data must be processed prior to model training.

- 2.2.1 Feature Selection. The data collection process using Wireshark yields a dataset containing seven distinct features shown in figure 2.
 - **Packet Number**: An integer representing the order in which the packets were captured during the Wireshark session. Provides insight on the order of the packets on the network interface.
 - **Time**: A floating point value representing the relative timestamp for each packet since the start of the Wireshark session. Similar to packet number, the time feature provides a better understanding of the packet order yet is more precise.
 - **Source**: A string value containing the internet protocol address of the device that the packet originated from
 - **Destination**: A string value containing the internet protocol address of the device that is receiving the packet.

- Protocol: A string value which contains the network protocol employed for the transmission of the packet.
 Three distinct protocols were identified within the datasets used in this project (QUIC, TCP and TLSv1.2).
- Length: An integer representing the size of a packet in bits
- Info: A string value containing more information pertaining to a packet. This value can vary heavily between packets.

As the objective of this project is to predict the origin of web traffic, both source and destination are excluded from the dataset used to train the models. Including these would be of course the simplest way to classify the data, of course, but the objective of this investigation is to have a highly applicable mechanism to classify network data when a small subset of information is available. Including the dependent variable of the research question as an independent variable would trivialize the research.

The info feature is also removed from our data, since its meaning highly varies from packet to packet. Two example Info fields are shown below:

Example 1:

"Protected Payload (KP0)"

Example 2:

"Initial, SCID=01d117623d73ec3a2fd3 9d62da73e7cc7e63ff11, PKN: 0, ACK"

There is no good reason to think this would help with classification, and even if it were to improve the accuracy with our test set, we have no ability to predict in good faith that this would always be the case, since there isn't a consistent semantic meaning.

Consequently, the features selected for training and testing the models are Time, Length, and Protocol. The Protocol feature is converted to an integer for compatibility with the models using a dictionary to map protocol names to a corresponding numerical representation (e.g., TCP: 0, QUIC: 1, TLSv1.2: 2) as shown in figure 3. Additionally, a categorical variable titled "Website" is created which displays which website the packet originated (e.g., LinkedIn, ChatGPT, Blackboard). The "Website" feature acts as the target variable and allows the models to detect relationships between the packets and their corresponding web traffic origin.

2.2.2 Data Formatting. To maintain normalized representation in all the collected datasets, a process of data harmonization was applied after collecting the necessary data. The dataset with the minimum number of observations is identified and accounted for when truncating the remaining two datasets. Subsequently, the three datasets are combined and rearranged to minimize bias while training each of our models. This process ensures a balanced yet unbiased dataset that

maximizes the effectiveness of our models. Figure 3 presents the refined data following the data preparation.

2.3 Employed models

For this project, we required a classification model that would be able to accurately predict the website of origin for a given packet. In the end, we chose to employ four different models: Logistic Regression K-Nearest Neighbors, Random Forest, and a Neural Network. The reason we selected more than one model was to be able to effectively compare the performance of each and determine which was the most effective for our solution. We tested the models sequentially, learning lessons along the way about the format of the data, and what similarities and differences exist in the data that could be potentially exploited by a subsequent model.

2.3.1 Logistic Regression (Baseline). As a starting point, we employed the Logistic Regression model as a baseline for our data analysis. Logistic regression is a statistical method used for binary classification tasks, where the goal is to predict the probability of an event happening or not happening. It was understood initially that this model may not be an ideal fit for our data; due to the regressions binary nature having more than two classifications is not ideal.

Characteristics of Logistic Regression:

- (1) **Binary Outcome**: Logistic regression is used for binary classification tasks, where the outcome variable has only two possible outcome
- (2) **Linear Relationship**: Logistic regression assumes a linear relationship between the independent variables and the log-odds of the outcome
- (3) **Probabilistic Output**: Instead of predicting discrete classes directly, logistic regression predicts the probability that a given observation belongs to a particular class
- (4) **Interpretability**: Logistic regression coefficients represent the change in the log-odds of the outcome for a one-unit change in the corresponding independent variable, making it easy to interpret the impact of each variable on the outcome. This characteristic was especially important for our baseline

The Logistic regression was an ideal choice for our baseline specifically because of its interpretability characteristics. We wanted to pick a model which would (1) allow us the swiftest entry point into our approach and (2) allow for us to not only have a simple foundation but a foundation from which we could improve upon.

The execution of this model was rather simple: splitting the data into training and testing sets and executing a standard regression to be analyzed. This simplistic approach, as previously mentioned, allowed for us to inspect which features were most advantageous for us and more importantly which features we would continue to analyze.

2.3.2 K-Nearest Neighbors. K-Nearest Neighbors (KNN) is a supervised learning model which classifies data points based on their proximity to a specified number (k) of existing data points.

The implementation process is as follows:

- (1) **Optimization and Training**: Prior to training the model, the optimal number of neighbors (k) for the model is identified using K-Fold cross validation. This technique involves dividing the training data into multiple folds and utilizing one fold for validation while the remaining ones are used for training. For every fold, a KNN model is trained with a specified k value and the accuracy (external validation) is calculated. Multiple different k values are tested using this process and the k value that yields the highest average external validation is used in the final model. After the optimal k value is calculated, the model achieved an accuracy of 97% on the training data.
- (2) Evaluating Model: In order to assess the effectiveness of the model, an independent dataset containing 2000 packets was collected and evaluated. Having the model evaluate a separate unseen dataset reveals a more realistic measurement of the accuracy of our model. After evaluating this dataset, the KNN model achieves an accuracy of 77%.
- 2.3.3 Random Forest. A Random Forest is a form of an ensemble model that is composed of multiple decision trees. Each tree is ran and gives its predicted classification, and the final classification is determined by a majority vote of all the trees (seen in figure 6).

Advantages of Random Forest Models:

- (1) **Ability to handle complex features**: Random forests have the ability to handle complex features, such as time and protocols in the case of network packets, and the relationships between them, which can be very important in classifying network traffic.
- (2) **High accuracy**: Random forests diminishes the problem of overfitting that can occur in decision trees, and can achieve higher accuracy, helping to more accurately determine website origins of a packet.
- (3) Feature importance: Random forests can provide insight into which features are most important in the classification process, such as protocol vs time in classifying websites for this project.
- (4) **Scalability**: Random forests can be scaled to handle much larger datasets, which can be very important

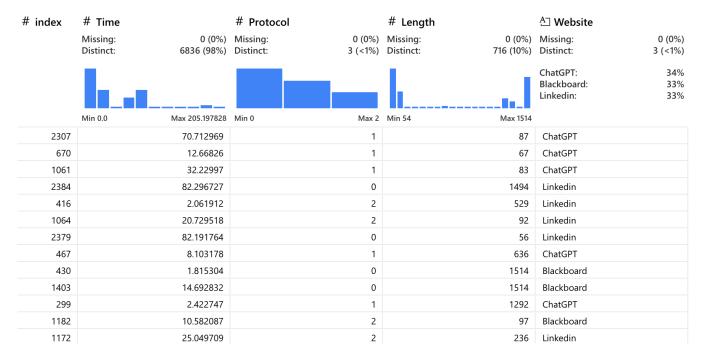


Figure 3: Processed Web Traffic Data

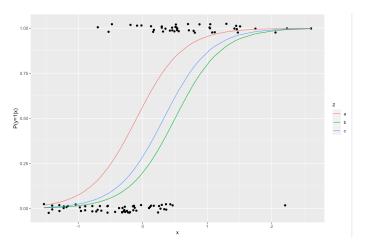


Figure 4: Logistic Regression model structure example

for network traffic which can receive a high volume of packet data extremely quick.

The library used for the Random Forest model was scikitlearn [3]. Sci-kit learn's implementation of Random Forests utilize aggregation to take a soft vote of the soft classifications (probability of the packet being from each class). Within soft voting, a weighted average of the probabilities from the classification are taken, and the class with the highest probability is chosen as the final classification.

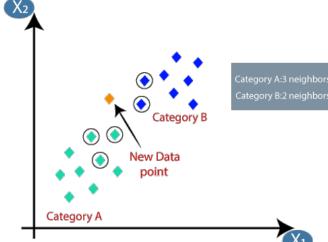


Figure 5: K-Nearest Neighbor model structure example

Within random forest models, many hyperparameters can be fine tuned to optimize the model. For this project, we chose to optimize the hyperparameters for n estimators (number of trees in the forest), max depth (maximum depth (levels) in the tree), and min samples split (minimum samples within a node to be able to split it into children). In order to optimize all of these hyperparameters, we used a grid search to

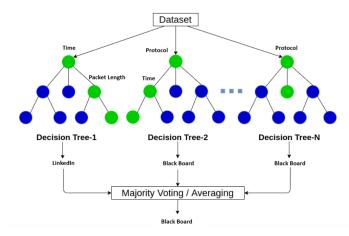


Figure 6: Random Forest model structure example

find the best combination of hyperparameters that would yield the highest accuracy. Within the grid search, we also employed the cross validation technique of Kfold validation, which divides the training data into k number of folds. This process is repeated k times, with each fold being used as the validation set exactly once. The results from each fold are then averaged to produce a single estimation of model performance. Overall, after running for many different combinations of values for the hyperparameters, the ones we decided to utilize, based on the highest test accuracy scores, were n estimators = 100, max depth = 13, and min samples split = 5.

2.3.4 Neural Network. The final model we chose to use was a Neural Network. A Neural Network is a machine learning model that tries to make decision as if it were a human brain. It consists of a graph of nodes, analogous to neurons, that are interconnected by directed edges, analogous to synapses.

The nodes are organized into layers, as specified here.

- (1) **Input Layer**: The input layer is, as it sounds, the interface through which the independent variable data comes into the model. In our case, since the model had three inputs, Time, Length, and Protocol, there were three nodes in the input layer.
- (2) Output Layer: The output layer is the layer that makes the decisions as to what classes the data are places in. In a "one-hot-encoding" scheme, which is what we employed, there are as many nodes as there are output classes in the data. For our purposes, this corresponds to three nodes, one for Blackboard, one for ChatGPT, and one for LinkedIn. One of the nodes is activated for each data point, and that is the class label it receives. In binary classification data sets, you could use a single node, and define a threshold, before which the output is classified as class 0 and after

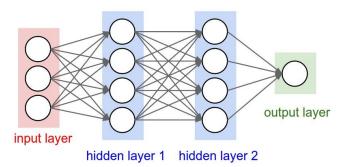


Figure 7: Neural Network Model Structure Example

which the output is classified as class 1. We could have done something similar, by having two thresholds, where either end of the distribution classifies into one of two classes, and the middle classifies into a third, but we felt that the one-hot method made more sense for a number of reasons. Namely, the activation functions available (touched on later) didn't provide adequate separation for the simplistic data we have, and the middle class being between two thresholds while the other two are open ended is an unfair and unnecessary bias.

(3) **Hidden Layer(s)**: The input and output layers can be directly connected by edges. This, if fully connected, would be called a perceptron. You'll recall our model, however, is called a multi-layer perceptron. This is because there are some *n* hidden layers, with *m* nodes each, connected to one-another between the input and output layers.

In a fully-connected neural network, or multilayer perceptron (MLP), which is what our model used, all of the nodes in each layer of the topology are connected to every node in the next layer, and every node in the previous layer. Nodes at the same layer are not connected.

The resulting graph structure is directed, meaning each edge moves in one direction, away from the input layer and towards the output layer. Figure 7 shows the topology and edge structure of an MLP with a 3 node input layer, like out model, a 1 node output layer, *unlike* our model, and two hidden layers with four nodes each.

Any given node is related to the connected nodes in the following way. Say node Y represents a node in the first hidden layer of Figure 7. There are 3 inputs from the input layer: x_1, x_2, x_3 , and four required outputs, all of which will be the same . Within the node, The three inputs are fed into this equation:

$$z = \sum_{i=1}^{3} (x_i \cdot w_i) + b$$

Variables w_i and b correspond to weight and bias, respectively. Both of these terms get set by the training process of the model, which we'll discuss shortly. The resulting value, z, is the input to an *activation function*, which, depending on which is used, modifies and normalizes the output before it is passed to the next node.

$$output = activation(z)$$

After activation, the math at this node is finished, and the post activation output becomes the input from this node for all the nodes in hidden layer 2.

Through a solving process, the specific algorithm of which (lbfgs) will be passed into the model, the weights and bias are all calculated for every node, in order to maximize accuracy.

These mathematical procedures would be heavily skewed by outliers in the dataset. Therefore, we made the choice to scale the data using ski-kit-learn's Standard Scaler class to scale all the data between 0 and 1.

Hyperparameters to a neural network define how it is built. We tuned these extensively, in multiple grid searches. The settled-upon hyperparameters include:

- Activation function: Logistic. The logistic, or sigmoid activation function shapes the input features into a binary output. This is necessary at each layer, as well as for one-hot encoding.
- (2) **Alpha (learning rate)**: $\alpha = 0.1$. This is an input to the Solver function, which helps to determine the rate at which the model learns.
- (3) Hidden Layer Sizes: 1 layer of 20 nodes.
- (4) Solver: Limited memory Broyden-Fletcher-Goldfarb-Shanno (lbfgs) algorithm. This is the quasi-Newton method by which the weight and bias hyperparameters are tuned.

As mentioned, these hyperparameters were not chosen at random. We ran a GridSearch on many possible combinations of parameters, exhaustively searching every combination. In order to do this, we used the GridSearchCV library in python from scikitlearn, which allowed us to divide the training data into 5 partitions. Then 5 separate times, each of the partitions could be used as the validation data, while the rest of the training data was used to solve the network. These 5 models were bulilt for every possible combination of hyperparameters, from the input hyperparameters shown below.

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After running this grid search, we calculated the validation accuracy from every model, and mean validation accuracy within the 5 folds for each set of hyperparameters. The highest validation score and therefore model we chose, was .94. Then, we rebuilt that network and scored it on the fully separate test data. We'll discuss results from the test data in the next section.

3 EVALUATION

This section comprehensively evaluates our models and their performance on the dataset and validating sets. Later, goes on to determine the model which best classified our selected websites based on network packet data.

3.1 Dataset

Our dataset of packets acquired from wireshark was split up in a couple of ways. First and foremost, we had completely separate training data and test data. The training data was used in training processes such as KFold validation that require it to be split "test"-train splits, but we refer to those "test" sets as validation sets, because they are part of the training process, and completely unrelated to the test data.

Moreover, we used the same test data on all four models, in order to ensure an accurate comparison between them, since their performance is therefore measured with similar constraints. We also used the same training data for all four models, except in the Neural Network we collected and added additional data. This is because Neural networks tend to need more data on which to train.

3.2 Evaluation Metrics

Accuracy and Macro F1 score were the primary metrics used to evaluate each model.

These metrics are applicable to all models and allow for a direct comparison for each of their performances. Two other metrics: precision and recall, help to provide a more detailed understanding of the model's performance on a class-to-class basis.

The equation for **Accuracy**:

$$Accuracy = \frac{TP + TN}{TS}$$

The equation for **Macro F1**:

$$MacroF1 = \frac{1}{n} \sum_{i=1}^{n} \frac{2 * Precision_{i} * Recall_{i}}{Precision_{i} + Recall_{i}}$$

The equation for **Recall**:

$$Recall = \frac{TP}{TP + FN}$$

The equation for **Precision**:

$$Accuracy = \frac{TP}{TP + FP}$$

Accuracy is the score that rates the effectiveness of a model at classification overall (ie. how many it predicted correctly out of the total set).

Macro F1 is the harmonic average of the of precision and recall for every class, and then taking the average of these calculated scores across all classes.

Recall is the calculation of how many the model correctly classified within a class (ie. how many truly belong in that class in the entire dataset). Precision is the calculation of how many the model correctly classified within a class (ie. can be thought of as class accuracy).

3.3 Model Comparison and Evaluation Results

The results in table 1 show that the neural network achieved the highest accuracy (85%) and Macro F1 score 84%. These results mean the model was the most effective at classifying the selected websites based on network packet data. This displays that the neural network is more effective at representing complex relationships in the data than the other models.

Table 1: Accuracy and Macro F1 Scores for Each Model

Model	Accuracy	Macro F1
Logistic Regression	56%	0.00
K-Nearest Neighbors	77%	0.00
Random Forest	83%	82%
Neural Network	85%	84%

The logistic regression has the lowest scores by far. This makes sense, since it was a dataset with 3 dependent variable classes, and logistic regression is a binary classifier. The model wasn't designed for this task, so a low accuracy makes sense.

include logreg scatterplot and actual scatterplot here As you can see from the scatterplot,

yap about scatter plots

Beyond the baseline logistic regression model, the other models all perform fairly similarly, in the high 70s and low

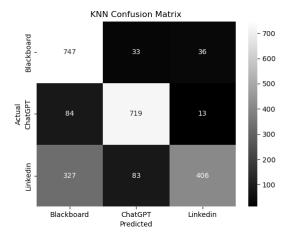


Figure 8: K Nearest Neighbors Model Heatmap

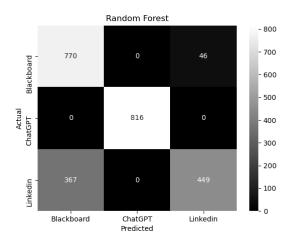


Figure 9: Random Forest Model Heatmap

80s in terms of test accuracy. These numbers are good, but not great, for a number of reasons.

First, as you can see from the Actual sample space scatter plot, there is a lot of clustering of the data, particularly in terms of the Protocol feature. ChatGPT uses almost exclusively QUIC protocols, whereas Blackboard and LinkedIn use a mix of TCP and TLSv1.2. This makes differentiating chatGPT from either of the others trivially easy, but disambiguating Blackboard and LinkedIn relatively hard.

You'll notice in the heatmaps in figures 8, 9, and 10 that chatGPT is classified 100% accurately by RF and NN, and is very close with KNN.

If all of our packets had their own protocols, we'd probably be able to use a simple decision tree with two splits and get nearly 100% accuracy. In reality, the physical closeness

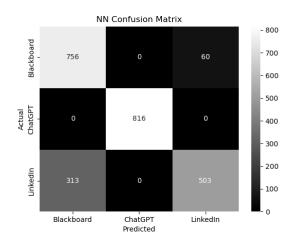


Figure 10: Neural Network Model Heatmap

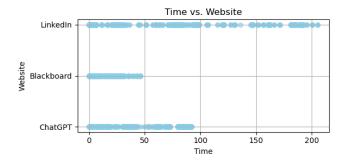


Figure 11: Scatter Plot of the Time feature

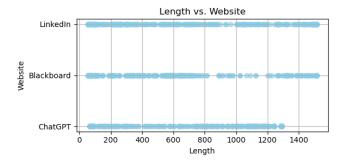


Figure 12: Scatter Plot of the Length feature

of LinkedIn and Blackboard data made it hard for KNN to accurately guess based on neighbors, for the Random Forest to build meaningfully effective trees, and for the Neural Network to solve itself cleanly.

Since Protocol wasn't deterministic at all for LinkedIn and Blackboard, the explanatory power that our model had must have come from Time and/or Length. In figure 11, see that

Blackboard has significantly lower time values that LinkedIn. This is likely where most of the explanatory power came from since length has a similar distribution between all websites, as per figure 12.

4 DISCUSSION & FUTURE WORK

4.1 Analysis of Results

4.2 Future Work

If we were to continue this research in the future, we'd want to explore some of the following areas.

First and foremost, we'd want to use a more meaningful variable instead of the unadjusted Time field as one of our input features. While it's true that time could be more meaningful if there was a set length of time in which each session of wireshark data, even controlling for that doesn't completely eliminate bias from the variable. A better approach would be to use time since previous packet, by getting the time field from the packet indexed before, and taking a difference, for every packet in the dataset (except the first, of course, which would have to be excluded). If we were to do this approach, there's potential that the variance of delays in packets, combined with the length variable which we'd still be measuring, would give our models information as to the amount of data transfer happening in the website, which could be explanatory towards its class.

Additionally, we would want to expand our dataset and collect data from more than three websites, with use cases across the Internet. Perheps if the model were trained on enough different types of sites, we'd be able to classify by website type, such as social media, shopping, etc. Then maybe the model would even be applicable to more domains that weren't even visited in the training data.

Finally, instead of looking statically at individual packets, we could use a flow of, say, 20 packets as input to our model. This would give us access to more independent variables, such as average length, average packet delay, etc. Since network communication is a constant flow of packets, it's possible that zooming out in this way from the way our model is currently would provide more explanatory power.

5 CONCLUSION

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