Biometric Authentication via Keystroke Sound

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Abstract

Unlike conventional one shot biometric authentication schemes, continuous authentication has a number of advantages, such as longer time for sensing, ability to rectify authentication decisions, and persistent verification of a user’s identity, which are critical in applications demanding enhanced security. However, conventional modalities, e.g., face, fingerprint, and keystroke dynamics, have various drawbacks in continuous authentication scenarios. In light of this, this paper proposes a novel non-intrusive and privacy-friendly biometric modality that utilizes keystroke sound. Given the keystroke sound recorded by a low-cost microphone, our system extracts discriminative features and performs matching between a gallery stream and a probe stream. Motivated by the concept of digraphs used in modeling keystroke dynamics, we learn a virtual alphabet from keystroke segments, from which the digraph latency within pairs of virtual letters, as well as other statistical features, are utilized by the score functions. The resultant multiple scores are indicative of the similarities between two sound streams, and are fused to make a final authentication decision. We collect a first-of-its-kind keystroke sound database of 45 subjects typing on a keyboard. Experiments on static text-based authentication, demonstrate the potential as well as limitations of this novel biometric modality.

1. Introduction

Biometric authentication is the process of verifying the identity of individuals by their physical characteristics, such as face [8] and fingerprint [16], or behavioral traits, such as gait and keystroke dynamics [2]. Most biometric authentication research has focused on one shot authentication where a subject’s identity is verified only once. However, one shot authentication has a number of drawbacks in many application scenarios. These include short sensing time, inability to rectify decisions, and enabled access for potentially unlimited periods of time [13]. In contrast, continuous authentication aims to continuously verify the identity of a subject over an extended period of time thereby ensuring that the integrity of an application is not compromised. It can not only address the aforementioned problems in one shot authentication, but can also substantially enhance the security of systems, such as computer user authentication in security-sensitive facilities.

Research on continuous authentication is relatively sparse, and researchers have explored a limited number of biometric modalities, such as face, fingerprint, keystroke dynamics, and mouse movement, for this purpose. However, each of these modalities has its own shortcomings. For example, face authentication demands uninterrupted sensing and monitoring of faces - an intrusive process from a user’s perspective. Similarly, the fingerprint sensor embedded on the mouse requires user collaboration in terms of precision in holding the mouse [16]. Although being non-intrusive, keystroke dynamics utilizes key-logging to record all typed texts and thus poses significant privacy risk in the case of surreptitious usage [14, 2].

To address these problems in continuous authentication, this paper presents a novel biometric modality based on keystroke acoustics. Nowadays, microphone has become a standard peripheral device that is embedded in PCs, monitors, and webcams. As shown in Figure 1 a microphone can be used to record the sound of a user typing on the keyboard, which is then input to the proposed system for...
Our technical approach to employ this novel biometric is largely motivated by prior work in keystroke dynamics. One of the most popular features used to model keystroke dynamics is digraph latency [10, 9, 3], which calculates the time difference between pressing the keys of two adjacent letters. It has been shown that the word-specific digraph is much more discriminative than the generic digraph, which is computed without regard to what letters were typed [15]. Considering that the acoustic signal from keystroke does not explicitly carry the information of what letter is typed, we propose a novel approach to employ the digraph feature by constructing a virtual alphabet. Given the acoustic signals from all training subjects, we first detect segments of keystrokes, whose mel-frequency cepstral coefficients (MFCC) [4] are fed into a K-means clustering routine. The resultant set of cluster centroids is considered as a virtual alphabet, which enables us to compute the most frequent digraphs (a pair of cluster centroids) and their latencies for each subject. Based upon the virtual alphabet, we also consider a number of other feature representation and scoring schemes. Eventually a score level fusion is employed to determine whether a probe stream matches with the gallery stream. In summary, this paper has three main contributions:

- We propose a novel keystroke sound-based biometric modality that offers non-intrusive, privacy-friendly, and potentially continuous authentication for computer users.
- We collect a first-of-its-kind sound database of users typing on a keyboard. The database and the experimental results are made publicly available so as to facilitate future research and comparison on this research topic.
- We propose a novel virtual alphabet-based approach to learn various score functions from acoustic signals, and a score-fusion approach to match two sound streams.

2. Prior Work

To the best of our knowledge, there is no prior work in exploring keystroke sound for the purpose of biometric authentication. As a result, we focus our literature survey on various biometric modalities for continuous authentication, and other applications of keystroke sound.

Face is one of the most popular modalities suggested for continuous user authentication, with the benefit of using existing cameras embedded in the monitor and no requirement for users’ cooperation [16, 12]. However, continuously capturing face images creates an intrusive and unfavorable computing environment for the user. Similarly, fingerprint has been suggested for continuous authentication by embedding a fingerprint sensor on a specific area of a mouse [16]. This can be intrusive since it substantially constrains the way a user operates a mouse. Mouse dynamics has been used as a biometric modality due to the distinctive characteristics exhibited in its movement when operated by a user [13]. However, as indicated in a recent paper [19], more than 3 minutes of mouse movement on average is required to make a reliable authentication decision, which can be a bottleneck when the user does not continuously use the mouse for an extended period of time.

Keystroke dynamics utilizes the habitual patterns and rhythms a user exhibits while typing on a keyboard. Although it has a long history dating back to telegraph in the 19th century and Morse Code in World War II, most of the prior work still focuses on static text [17, 20, 21], i.e., all users...
type the same text. Only a few efforts have addressed the issue of free text (i.e., a user can type arbitrary text) which is necessary for continuous authentication. Nevertheless, one major drawback of keystroke dynamics is the fact that it poses significant privacy risk to the user because all typed texts can be potentially recorded via key logging.

As far as recorded keystroke sound is concerned, there has been a series of work on acoustic cryptanalysis in the computer security community. The focus has been on designing advanced signal processing and machine learning algorithms to recover the typed letters from the keystroke sound. One of their main assumptions is that when pressed, each key/letter will emit a slightly different acoustic signal, which motivates us to learn a virtual alphabet for our biometrics authentication application. Assuming the success of this line of work in the future, we may leverage it to combine the best worlds of both sides, the keystroke dynamics based authentication using recovered keys and the enhanced discrimination due to additional 1D acoustic signals over step-function-like simple key logging signals.

3. Our Algorithm

As a biometric modality, our algorithm seeks to calculate a similarity score between a probe sound stream recorded during the authentication stage and a gallery sound stream, whose features have been pre-computed and saved as a biometric template during the enrollment stage. Both sound streams are recorded when a user is typing on a keyboard with minimal background noise, and they are described in detail in Section 4.1. In this section, we present our algorithm for calculating the similarity score between two sound streams.

Figure 3. A raw acoustic signal containing one keystroke.

Figure 4. Temporal segmentation by thresholding the FFT power.

For an acoustic signal of typing $g(t)$ composed of keystrokes and silent periods, the first task before feature extraction is to identify portions in the stream where a keystroke occurs since it is the keystroke that carries the discriminative information about an individual typist. As shown in Figure 3, a keystroke is composed of key pressing and key release, where the former comprises of the sound emitted when a finger touches the surface of a key as well as the sound due to the key being pressed down. We denote a keystroke as $k_i$, with a start time of $t_i$ and a duration of $L$. In our algorithm, $L$ is set to be 40 ms, since it covers the full length of most keystrokes.

Motivated by [21], we conduct the temporal segmentation based on the observation that the energy of keystroke is concentrated in the frequencies between 400 Hz and 12K Hz, while the background noise occupies other frequency ranges. We compute the 5-ms windowed Fast Fourier Transform (FFT) of an acoustic signal using a sliding window approach, where the magnitudes of outputs in the range of 400 Hz and 12K Hz are summarized to produce an aggregate curve of the FFT power. By setting a threshold on the curve, we identify the start of keystrokes, as shown in Figure 4. Ideally this temporal segmentation should detect all keystrokes, instead of the background noise. Our current simple threshold-based method may not achieve this reliably as yet, and in the future we will investigate more advanced methods such as adaptive thresholding or supervised learning approaches.
Once the start of a keystroke \( t_i \) is determined, we convert the acoustic signal within a keystroke segment, \( q(t_i, \ldots, t_i + L) \), to a feature representation \( f_i \) for future processing. The standard MFCC feature are utilized due to its wide applications in speech recognition and demonstrated effectiveness in key recovery [21]. Our MFCC implementation uses the same parameter setup as [21]. That is, we have 32 channels of the Mel-Scale Filter Bank and use 16 coefficients and 32 filters with 10-ms windows that are shifted by 2.5 ms. The resultant feature of a keystroke \( k_i \) is a 256-dim vector \( f_i \).

3.2. Virtual Alphabet via Clustering

Most of the prior work in keystroke dynamics uses digraphs statistics, which is the mean and standard deviation of delays between two individual keys or two groups of keys, or trigrams, which is the delay across three keys, to model keystroke dynamics. It has also been shown that word-specific digraph, which is computed on the same two keys, is much more discriminative than the generic digraph [13]. In keystroke dynamics, such word information is readily available since key logging records the letter associated with each keystroke. However, this is not the case in our keystroke acoustic signal. Hence, in order to enjoy the benefit of digraph in our application, we strive to achieve two goals from the acoustic signal:

- Estimate the starting time of each keystroke;
- Infer or label the letter pressed at each keystroke.

While the first goal is addressed by the temporal segmentation process, the second goal requires knowledge of the pressed key in order to construct word-specific digraphs. However, as shown in acoustic cryptanalysis [6], precisely recognizing the typed key itself is an ongoing research topic. Hence, we take a different approach by aiming to associate each keystroke to a unique label, with the hope that different physical keys will correspond to different labels, and different typists will generate different labels when pressing the same key. We call each label a virtual letter and the collection of which is called a virtual alphabet. The learning of the virtual alphabet is accomplished by applying K-means clustering to the MFCC features of all keystrokes in the training set.

Assuming the training set includes streams from \( J \) subjects, each with \( I_j \) keystrokes, the input of the K-means is \( \{f_i^j\} \), where \( j \in [1, J] \) and \( i \in [1, I_j] \). The K-means clustering partitions the input samples into \( K \) clusters, with centroids \( \bar{m}_j(k) \). We set \( K \) to be 30 considering that the total number of common keys on a keyboard is around 30.

During all three stages, given an original acoustic signal represented as a collection of keystrokes \( \mathbf{K} = \{k_i\} \), we compute the Euclidean distance between the MFCC feature of a keystroke \( f_i \) and each of the cluster centroids. The centroid with the minimal distance will assign its index to the keystroke as its corresponding virtual letter, i.e., \( l_i = k \). Thus, in our algorithm a virtual letter is simply a number between 1 and \( K \). Finally, we can represent a keystroke as a triplet \( k_i = \{t_i, f_i, l_i\} \).

3.3. Score Functions

Given the aforementioned feature representation scheme, we next investigate a set of score functions to compute the similarity measure between two sets of features, listed as follows.

**Digraph statistic** As a representation of individual typing characteristics, digraph statistic refers to the statistics of the latency, in terms of mean and standard deviation, within frequent pairs of letters. A pair of letters is named digraph, with examples such as \((t, h), (h, e)\). In our algorithm, a virtual alphabet with \( K \) letters can generate \( K^2 \) digraphs, such as \((2, 5), (7, 1)\). During the training stage, we count the frequency of each digraph by passing through adjacent keystrokes, \( l_{i-1} \) and \( l_i \), in the entire training set. Then we generate a list of top \( N \) most frequent digraphs, each denoted as \( d_n = \{k_{n1}, k_{n2}\} \). Given the \( \{k_i\} \) representation of an acoustic signal, for each one of \( N \) digraphs, we then compute the mean, \( \{m_n\} \), and the standard deviation, \( \{\sigma_n\} \), of the time difference variable \( \Delta t = t_i - t_{i-1} \) where \( l_i = k_{n2} \) and \( l_{i-1} = k_{n1} \). Finally, the similarity score between two arbitrary length signals, \( \mathbf{K} \) and \( \mathbf{K}' \), is computed using the following equation:

\[
S_i(\mathbf{K}, \mathbf{K}') = \sum_{n=1}^{N} w^n_i \left[ \sum_{\Delta t} \sqrt{N(\Delta t; m_n, \sigma^2_n)}N(\Delta t; m'_n, \sigma^2'_n) \right],
\]

which basically summarizes the overlapping region between two Gaussian distributions of the same digraph, weighted by \( w^n_i \). The overlapping region is computed via the Bhattacharyya coefficient, and \( w^n_i \) is the overall frequency of each digraph learned from the training set. We set \( N \) to be 50 in our algorithm.

**Histogram of digraphs** In our virtual alphabet represent-
tation, different subjects may produce different virtual letters when pressing the same key. This implies that different subjects are likely to generate different digraphs when typing the same word. Hence, it is expected that we can use the frequencies of popular digraphs as a cue for autotyping the same word. Hence, it is expected that we can

differ when pressing the same key. This implies that different subjects may produce different virtual letters. It can be seen that 1) there is distinct inter-letter separation among virtual letters; 2) within each virtual letter, there is intra-letter separation due to individuality. Hence, we would like to utilize this intra-letter separation in our score function computation. For an acoustic signal, we compute the mean of $f_i$ associated with each virtual letter, which results in $K = 30$ mean MFCC features, as follows:

\begin{equation}
S_2(K, K') = h^T h'.
\end{equation}

**Intra-letter distance** We use the virtual letter to represent similar keystrokes from various key pressing of various subjects. Hence, within a virtual letter, it is very likely that different subjects will have different distributions. Figure [5] provides evidences for this observation by showing the mean MFCC features of 20 training subjects within virtual letters. It can be seen that 1) there is distinct inter-letter separation among virtual letters; 2) within each virtual letter, there is intra-letter separation due to individuality. Hence, we would like to utilize this intra-letter separation in our score function computation. For an acoustic signal, we compute the mean of $f_i$ associated with each virtual letter, which results in $K = 30$ mean MFCC features, as follows:

\begin{equation}
S_2(K, K') = h^T h'.
\end{equation}

**Algorithm 1:** Feature extraction algorithm.

**Algorithm 2:** Algorithm in the authentication stage.

Given two acoustic signals $K$ and $K'$, we use Equation (5) to compute the Euclidean distance between the corresponding means and sum up by using a weight $w_k^2$, which is the overall frequency of each virtual letter among all keystroke segments and is computed from the training set. The sign $-1$ is to make sure that on average the genuine probe has a larger score than the impostor probe.

\begin{equation}
S_3(K, K') = - \sum_{k=1}^{K} w_k^2 ||\bar{f}_k - \bar{f}'_k||_2.
\end{equation}

So far we have constructed a new feature set for one acoustic signal, denoted as $\{m_n, \sigma_n, h_n, \bar{f}_k\}$ where $n \in [1, N]$ and $k \in [1, K]$. We summarize the feature extraction algorithm in Algorithm 1. If the acoustic signal is from the gallery stream, we call the resultant feature set as a biometric template of the gallery subject, which is computed during enrollment and stored for future matching with a probe stream.

**Score fusion** Once the three scores are computed, we fuse them to generate one value to determine whether the authentication claim should be accepted or rejected. In our system we use a simple score-level fusion where the three normalized scores are added to create the overall similarity score between two streams,

\begin{equation}
S = \sum_{v=1}^{3} \frac{S_v - m_{sv}}{\sigma_{sv}}.
\end{equation}

The score normalization is conducted by using the mean $m_{sv}$ and standard deviation $\sigma_{sv}$ of the score distribution learned from the training data, such that the normalized scores fall in a standard normal distribution. We summarize the algorithm during the authentication stage in Algorithm 2. In the future, more advanced fusion mechanisms
4. Experiments

In this section, we present an overview of the database collected for this experiment which can be used for other typing based biometrics. We also present our specific experimental setup and the results of our static-text experiments.

4.1. Keystroke Sound Database

Since keystroke sound is a novel biometric modality without any prior database, we develop a capture protocol that ensures the collected database is not only useful for the current research problem, but also beneficial to other researchers on the general topic of typing-based biometrics.

While developing the protocol, our first goal is to study whether the keystroke sound should be based on static text or free text, i.e., does a subject have to type the same text to be authenticated? The same question was addressed in keystroke dynamics, where there is substantially more research effort on static text than free text [2]. To answer this question, we request each subject to perform typing in two sessions. In the first session, a subject types one paragraph four times (the first paragraph of “A Tale of Two Cities” by Charles Dickens displayed on the monitor), with a 2–3 seconds break between trials. Multiple typing instances of the same paragraph enable the study of static text based authentication. In the second session, a subject types a half-page letter with any content to his/her family. It can be seen from this session that most users make spontaneous pauses during typing, which mimics well the real-world typing scenario. Normally one subject spends around 5–8 minutes on each session, depending on the typing speed.

A second consideration is the recording environment. The background environment plays an important role in the usability of the data. Low frequency pitches from computers, heaters, and lights can create distractions to the sound of the actual typing. The placement of the sensor relative to the keyboard as well as in the room can also introduce differences due to echoes. All subjects in the study performed the typing with the same computer, keyboard, and microphone, in the same room, under reasonably similar conditions. Care was taken to use non-verbal communication during the experiments to eliminate human speech from corrupting the audio stream. Nevertheless, standard workplace noises still exist in the background, e.g. doors opening, chairs rolling, and people walking.

The third consideration is the hardware setup. We use a US standard QWERTY keyboard for data collection. Although there are many available options for microphones, we decide to utilize an inexpensive webcam with embedded microphone, which is centered on the top of the monitor and pointed toward the keyboard. This setup allows us to capture both the video of hand movement and the audio recording of keyboard typing. Thus, a multi-modal (visual and acoustic) typing-based continuous authentication system can be developed in the future based on this database. The sound is captured at 48,000 Hz with a single channel.

Thus far our database contains 45 subjects with different years of experience in using the keyboard. The database is formed from students and faculties at Michigan State University, whose distributions of typing experience and age are displayed in Figure 6 and Table 1. The database of keystroke sound is available at [http://www.cse.msu.edu/~liuxm/typing](http://www.cse.msu.edu/~liuxm/typing) and is intended to facilitate future research, discussion, and performance comparison on this topic.

4.2. Results

We use the static portion of the database for our experiments and leave the use of the free text for future work. Our algorithm requires a separate training dataset for the purpose of learning a virtual alphabet, top digraphs, and the statistics of score distributions. Hence, we randomly partition the database into 20 subjects for training and the remaining 25 subjects for testing our algorithm.

For each subject, we use the first typing trial as the gallery stream and portions of the fourth typing trial as the probe stream. We choose the first and last paragraph to maximize the intra-subject differences for the static text. For the probe streams, we need to balance two concerns. Firstly, we want to use as many probes as possible to enhance the statistical significance of our experiments, which requires that we use a shorter partial paragraph. Secondly, we want to use longer probe streams to accurately calculate features for a given subject. We decide to form 7 continuous probe streams for each user by using 70% of the fourth paragraph starting at the 0%, 5%, 10%, 15%, 20%, 25%, and 30% mark of the paragraph. This overlapping nature allows us to balance both concerns, while also simulates a continuous environment where the algorithm works with an incoming data stream and the results of
the current probe build on the results of the previous probes. The average probe length is 62 seconds. The total number of training probe streams is 140 with 3500 training cases. The total number of testing probe streams is 175 with 4375 testing cases.

**Evaluation Metrics** We use the standard biometrics authentication metric, the ROC curve, to measure performance. The ROC curve has two axes, False Positive Rate (FPR), the fraction of impostor pairs incorrectly deemed to be genuine, and True Positive Rate (TPR), the fraction of genuine pairs correctly deemed to be genuine. We use the Equal Error Rate (EER) to succinctly summarize the ROC curve. We also use the probability distributions of the genuine and impostor scores to study the performance.

**Feature Performance Comparison** Figure 7 illustrates the separability of the three score functions and the overall fused score on the testing data. Figure 8 displays the same information in the standard ROC format. We can make a number of observations. Firstly, the individual score function distributions all display overlap between the genuine and impostor probes. The task of identifying a single feature representation scheme to authenticate users via keystroke sounds proves challenging. Digraph statistics, histogram of digraphs, and intra-letter distance provide 31%, 32%, and 33% EER, respectively. Secondly, despite the overlap, there is still some separation between the genuine and impostor probes. We see Gaussian-like distribution for digraph statistics and intra-letter distance, which implies that in real-world applications with unseen data, the proposed score functions will be useful. The histogram of digraphs displays a bimodal Gaussian distribution for the genuine class, which indicates that it will behave on the extremes of either providing very useful information or limited discriminability. Furthermore, by using fusion, we create a new fused score, which produces better results and indicates that the individual score functions capture different aspects of the subject typing behavior. The result with the fused score has an EER of 25%.

**Computation Efficiency** Since a probe stream is an one-dimensional signal, our algorithm can operate in real time, with very minimal CPU load, which is a very favorable property for continuous authentication. For a 60-second probe stream, it takes approximately 20 seconds to create the feature representation with the majority of the time spent identifying the locations of keystrokes. Once the features have been extracted, it takes less than 0.1 seconds to compute the score functions against a biometric template. A future work is to design an incremental score function. This is important since we would like to perform authentication in the online mode, as the sound stream is continuously received.

**5. Conclusions**

In this paper, we presented keystroke sound as a potential biometric for use in continuous authentication systems. The proposed biometric is non-intrusive and privacy-friendly. It can be easily implemented with standard computer hardware and requires minimal computational overhead. To facilitate this research, we collected both acoustic and visual typing data from 45 individuals. The database size is on par with the well-know CMU benchmark keystroke database consisting of 51 subjects. We designed multiple approaches to compute match scores between a gallery and probe keystroke acoustic stream. In particular, we proposed a fusion of digraph statistics, histogram of digraphs, and intra-letter distances to authenticate a user. We obtained an initial EER of 25% on a database of 45 subjects. This
suggests that more sophisticated features have to be investigated in the future. While these are preliminary results, with the accuracy far below well-researched biometric modalities such as face or fingerprint, we wish to emphasize that the intent of this feasibility study is to open up a new line of exciting research on typing-based biometrics. We anticipate other interested researchers to commence working on keystroke acoustics as a biometric modality - either by itself or in conjunction with other modalities - in continuous authentication applications.

Our future work on this topic will cover the following aspects. First, we will increase the subject size of our database to provide more training samples. We will continue to collect data from a larger number of users as well as more samples from the current users to help obtain a better understanding of the inter and intra-class variability of this biometric. Second, we will explore methods for better identifying keystrokes through supervised learning or context-sensitive thresholding. This will allow for more robust authentication in the presence of background noise typical of a normal work environment. Third, we will explore the changes that occur when comparing free text, by utilizing the second session of our database. Finally, we will investigate new feature representation schemes to address the intra-class variability of this biometric.

It is interesting to note a potential attack on this biometric system which consists of recording a genuine user’s typing sounds and then replaying the recording while the impostor is using the computer in a mainly silent manner through use of only the mouse. A means of foiling this attack is to log only the times of keystrokes and ensure alignment of the detected keystroke times from the audio and the actual keystroke times. It is also desirable to completely fuse this method with standard keystroke dynamics methods where the audio and virtual digraphs add extra discriminating information to the physical digraphs.

References


