

Cost-constrained Drone Presence Detection through Smart Sound Processing

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Abstract: Sometimes, drones lead to problems of invasion of privacy or access to restricted areas. Because of that, it is important to develop a system capable of detecting the presence of these vehicles in real time in environments where they could be used for malicious purposes. However, the computational cost associated to that system must be limited if it has to work in an autonomous way. In this manuscript an algorithm based on Smart Sound Processing techniques has been developed. Feature extraction, cost constrained feature selection and detection processes, typically implemented in pattern recognition systems, are applied. Results show that it is possible to detect the presence of drones with low cost feature subsets, where MFCCs and pitch are the most relevant ones.

1 INTRODUCTION

The use of Unmanned Aerial Vehicles, also known as drones, is on the rise in the society, mainly because of the advantages they offer. However, these vehicles usually run into problems of invasion of privacy or access to hazardous areas (e.g. airports). For this reason it is important to develop a system capable of detecting the presence of drones in particular environments where they could be used for malicious purposes, such as households, public buildings or restricted-access areas. In the state of the art there are many studies which deal with this issue, trying to detect and locate drones (Ganti and Kim, 2016). The wide range of methods includes audio, video, temperature, radar and radio frequency based detection.

Video detection systems can cover long distances, but there is a difficulty when distinguishing between drones and birds, even after including bird flight patterns which drones do not follow (Ganti and Kim, 2016). In addition, the computational cost of this kind of systems is high. Talking about the temperature-based detection, it is an efficient solution if the drone uses a propulsion engine, which usually appears in fixed-wing drones. However, most current drones are made of plastic and their electric engines do not radiate much heat.

Systems based on radar signal are useful for air-

craft detection, but the small size of the drones complicates their detection. Some manuscripts are working on this alternative (Drozdowicz et al., 2016). Related to radio frequency based methods, they are useful for the problem at hand since radio frequency is the communication mode used between drones and the remote controller (Nguyen et al., 2016). However, the use of Wi-Fi range (2.4-5 GHz) in no-license channels causes the appearance of high interferences.

Some proposals have based their study on audio information, mixed or not with video one. Some authors propose the use of an array of microphones and an infrared camera to get the information (Case et al., 2008). They try to trace the path followed by the drone through beamforming techniques. Others use only one microphone, but they are focusing on detecting a particular model of drone, so the results could not be generalizable (King and Faruque, 2016). In one manuscript, the authors analyze video information to detect the difference between frames, and in this way they track the drone movement, while they use audio information for detecting the vehicle with a threshold in frequency (Ganti and Kim, 2016). The problem is that it is not very effective when background noise is high. In addition, audio appears to be more reliable for detecting drones according to some studies (Liu et al., 2017).

This manuscript proposes a real-time implemen-

tation of an energy-efficient system capable of detecting drone presence in smart environments. We want the system to work in an autonomous way, so computational cost related to the clock frequency of the processing units will be strictly constrained. In this sense, evolutionary computation (i.e. genetic algorithms) is proposed for selecting a reduced set of features from the full set calculated previously, allowing a good tradeoff between performance and computational cost.

2 SMART SOUND PROCESSING (SSP) SYSTEM

In order to detect drone presence, our study will be based on an efficient system successfully used in other applications, like violence detection (Bautista-Durán et al., 2017). This is because this set includes features like pitch, which can be useful for detecting the frequency associated to the drone engine, as well as the rotation speed, size and material of the propellers. The system has the objective of studying solutions for audio-based drone detection in real environments and in real time, where the system has to make a decision every T seconds. Fig. 1 shows the system diagram, whose steps will be explained in the following sections.

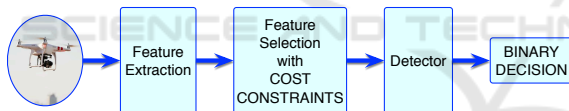


Figure 1: Scheme of the system.

2.1 Feature Extraction

The objective of this step is to extract useful information from the audio signal in the form of features. There are several audio features that have demonstrated to be really useful in other applications, fundamentally related to speech problems (Giannakopoulos et al., 2006; Mohino et al., 2011; Gil-Pita et al., 2015). In this manuscript we will apply this type of features to the problem of drone detection. In this section a theoretical description of the features will be made. To extract the features, the audio segments of T seconds are divided into M frames of L samples with an overlap of $S\%$. The following features have been considered:

- The Mel-Frequency Cepstral Coefficients (MFCCs). They are N parameters calculated from the spectrum that are typically used for

speech recognition. With this measurement, a compact representation of the spectral envelope is obtained. The objective is to emulate the human ear non-linear frequency response through a set of filters on non-linearly spaced frequency bands (Gil-Pita et al., 2015).

- The Delta Mel-Frequency Cepstral Coefficients (Δ MFCCs). They are calculated differentiating the previous MFCCs in two different time frames.
- The Pitch. This feature is related to the fundamental frequency and determines the tone of the speech. It allows to distinguish a person from another. In this manuscript the pitch is evaluated in every frame through the autocorrelation of the error of a linear predictor with P coefficients (Mohino et al., 2011).
- The Harmonic Noise Rate (HNR). With this feature it is feasible to evaluate the purity of the speech. It measures the relation between the harmonic energy produced by the vocal cords and the non-harmonic energy.
- The Ratio of Unvoiced time Frames (RUF). It measures the presence or absence of clear or strong speech. The computation consists of dividing the number of time frames with detected pitch by the total number of frames.
- The Short Time Energy (STE), which is the energy of the short speech segment. It is a simple and effective parameter for both voiced and unvoiced frames (Jalil et al., 2013).
- The Energy Entropy (EE). It allows to detect changes in the energy level of the audio. It is useful for detecting a quick emergence of a drone in the environment due to rapid changes in the energy of the audio. To evaluate this measurement, each time frame is divided into B blocks, and the energy of each block is then measured.
- The Zero Crossing Rate (ZCR). It is one of the most used audio features in time domain. To calculate it, the number of sign changes is divided by the total length of the frame.
- The Spectral Rolloff (SR). It is calculated in the frequency domain and is defined as the frequency below which $c\%$ of the magnitude distribution of Short Time Fourier Transform (STFT) coefficients are concentrated for a frame.
- The Spectral Centroid (SC) is the center of gravity of the magnitude spectrum of the STFT.
- The Spectral Flux (SF) measures the spectral changes between successive frames.

Once these features have been extracted, some statistics are applied to them (the mean and the standard deviation).

2.2 Feature Selection with Cost Constraints

If we want to get an energy-efficient real time system for detecting drone presence, it will have the restriction of consumption, as it will be implemented in some place to work in an autonomous way. In this scenario, computational cost is an important aspect to consider. In order to calculate the computational cost of our system, we have computed the resources that each feature requires determining the number of Floating Point Operations Per Second (FLOPS) (Qian, 2015), which is directly related to the power consumption of the device. The number of FLOPS of the system will depend on the set of selected features, so it must be taken into account which ones are used in each case (Bautista-Durán et al., 2017).

Thus, the cost of each feature has been evaluated and some equations are proposed with the objective of generalizing the cost according to some parameters that will be explained. As stated above, the feature extraction process splits the audio frame of $N_{samples}$ (so that $T = N_{samples}/f_s$, being f_s the sampling frequency) into M frames of L samples, with an overlap between them of $S\%$, so that:

$$M = \left\lfloor \frac{N_{samples}}{S \cdot L} \right\rfloor \quad (1)$$

Some aspects must be taken into account for the analysis. First of all, some features will have more impact in cost than others (e.g. MFCCs or pitch-based ones). In addition, some features need to apply the same processing blocks, so their computation do not have to be repeated. Considering the measurements of Section 2.1, four processing blocks that are shared along more than one measurement have been identified:

- The STFT is shared by the MFCCs, the Δ MFCCs, the SR, the SC and the SF.
- The MFCCs are shared by the MFCCs and the Δ MFCCs.
- The pitch is shared by the HNR and the RUF.
- The energy is shared by the STE and the EE.

In Table 1 the four processing blocks and their equations are shown. Four binary variables b_1 , b_2 , b_3 and b_4 related to B_1 , B_2 , B_3 and B_4 (the number of operations associated to the previous processing blocks) will be defined to determine if the set of features selected requires or not the evaluation of these blocks.

Thus, the total cost C will be calculated using Equation (2).

$$C = \sum_{i=1}^4 b_i \cdot B_i + \sum_{j=1}^{11} s_j \cdot C_j, \quad (2)$$

where C_j is the additional cost of each feature and s_j is a binary value which indicates if the feature is selected or not. Taking into account that the proposed system makes a decision every T seconds, the FLOPS can be evaluated.

As there are some features which are linked and depend on others, we have grouped the measurements into 8 groups: G_1 (including MFCCs and Δ MFCCs), G_2 (including Pitch, HNR, and RUF), G_3 (STE), G_4 (EE), G_5 (ZCR), G_6 (SR), G_7 (SC) and G_8 (SF). The groups, number of features of each measurement, values of b_S , b_M , b_P and b_E , and the equations of additional cost C_f associated to each measurement are detailed in Table 2. There we can see a typical cost of the problem at hand, considering each feature is selected individually, so the shared blocks need to be computed in each of them. The parameters used for solving the equations are: $B = 10$ blocks, $L = 512$ samples, $M = 31$ frames, $N = 25$ MFCCs coefficients, $P = 10$ Levinson coefficients and $S = 50\%$ overlap.

As it has been discussed, it is necessary to find a reduced set from the 117 features that allows obtaining a good performance and controlling the computational cost of the system. For this purpose, evolutionary algorithms have been implemented in the manuscript (Haupt et al., 1998). The configuration of this algorithm includes the next parameters: 100 individuals, 10 parents, 90 regenerated sons, percentage of mutation of 2%, 30 generations, 10 repetitions of the whole algorithm and minimization of the error rate as adaptive function.

2.3 Detectors

To evaluate the results and make a decision about the presence of drone sound, a detector has to be applied. In the present case, two different detectors have been used: the Least Squares Linear Discriminant (LSLD) and a reduced version of the Least Squares Quadratic Discriminant (LSQD). The computation of the two detectors is shown in Equations 3 and 4. (García-Gómez et al., 2016). They are obtained using the Wiener-Hopf equations. (Van Trees, 2004)

$$y = w_0 + \sum_{n=1}^L w_n x_n, \quad (3)$$

$$y = w_0 + \sum_{n=1}^L w_n x_n \sum_{m=1}^L \sum_{m=1}^n x_m x_n v_{mn}, \quad (4)$$

Table 1: Cost of the shared processing blocks.

Block	Cost of the block (No. operations)
STFT	$B_1 = L(M-1)(5\log_2 L + 2) + 4L + 15$
MFCCs	$B_2 = (L \cdot S + 1)(M(2N + 5) + 10N + 23) + N(3N + 11) + N \cdot M(2N + 7) + 29$
Pitch	$B_3 = 2L \cdot M(5\log_2 L + P + 3) + M(P(2P^2 + P + 2L + 1) - L) + 1$
Energy	$B_4 = M(2L + 3) - 4$

Table 2: Details of the groups of features.

Group	Caract	No. feats	b_1	b_2	b_3	b_4	Additional cost (No. operations)	Typical cost (MFLOPS)
G_1	MFCCs	50	1	1	0	0	$C_1 = 0$	1.25
	Δ MFCCs	50	1	1	0	0	$C_2 = N(M-2) + 1$	1.26
G_2	Pitch	2	0	0	1	0	$C_3 = 0$	2.21
	HNR	2	0	0	1	0	$C_4 = 9M$	2.21
	RUF	1	0	0	1	0	$C_5 = M$	2.21
G_3	STE	2	0	0	0	1	$C_6 = 0$	0.03
G_4	EE	2	0	0	0	1	$C_7 = M([2L/B] + 3B - 5) + 6B + 3$	0.06
G_5	ZCR	2	0	0	0	0	$C_8 = (6M + 1)(L - 1)$	0.10
G_6	SR	2	1	0	0	0	$C_9 = M(5N + 8) + 2[M(L \cdot S - 1)/3]$	0.74
G_7	SC	2	1	0	0	0	$C_{10} = M(8N + L \cdot S + 6) + L \cdot S + 4$	0.75
G_8	SF	2	1	0	0	0	$C_{11} = M(9N + 5) - 3N + 1$	0.74

where x_n and x_m are the training patterns, w_n and v_{mn} are the weights associated to them, w_0 is a bias term and y is the combination of the training patterns. A threshold will be applied to this combination to obtain the binary decision about drone presence.

It is important to indicate that in the beginning more complex detectors were considered (e.g. artificial neural networks). However they were discarded because the results were not as good as expected, due to the fact that overtraining problems appear as the dataset is not large enough.

3 RESULTS

To validate the system we have carried out some experiments using a dataset of audio files. These audio files have been divided in segments of $T = 1$ second, which indicates how often a decision is made. All the files have been resampled to a sampling frequency of $f_s = 8,000$ Hz. Each frame is divided in windows of $L = 512$ length and $S = 50\%$ overlap between windows, resulting in a total of $M = 31$ frames per segment. Then steps detailed in previous sections have been followed, including feature extraction, feature selection and detection.

The algorithm has been applied using a constraint related to the computational cost. Some cost thresholds measured in ‘‘Maximum number of Mega Floating Operations Per Second’’ (MaxMFLOPS) have been applied (0.5, 1, 1.5, 2, 2.5, 3, 3.5 and 4 MaxMFLOPS). This means that the sum of costs of the selected features has to be below these values. The up-

per limit is never reached, since the cost associated to the case of selecting all the features is below 4 MaxMFLOPS. Once the best features have been selected, a trained detector makes the final decision.

The datasets used in the state of the art are not suitable for our problem for several reasons: they just include a model of drone, or the environmental conditions do not change. Because of that, we have used a novel dataset that was developed in a previous work (García-Gomez et al., 2017). In this dataset, drones in motion and in a static position are included, as well as different models of them (Cheerson CX10, DJI Phantom 3, Eachine Racer 250, etc.). In order to make the database more challenging, similar no-drone sounds are included too (plane, helicopter, mower, etc.). The main characteristics of the used database are: total duration of 3671 seconds, duration of drone sound of 1913 seconds, percentage of drone presence of 50.08%, 36 fragments, minimum audio length of 6 seconds and maximum audio length of 316 seconds. More details about the dataset can be found in (García-Gomez et al., 2017).

The method of validation implemented has been a tailored version of k -fold cross-validation, since it allows avoiding loss of generalization of the results. The data is divided in k subsets, so that each subset is used for testing and the remaining $k - 1$ are used for training. In the case at hand, 36 folds with different size have been used, each fold containing a different audio file. In that way, we ensure that data from the same model of drone or with the same environmental conditions are not used both for training and testing at the same time.

3.1 Analysis of the Computational Cost Constraints

Now we will evaluate the effect of the limits in the computational cost available, as well as the groups of features more selected and useful. Table 3 displays the error rate and the percentages of appearance (selection rates) of the groups, in function of the maximum cost established in MFLOPS, using the LSLD. The error rate is the sum of the decisions where the system says there is drone presence and it fails because there is no drone in the environment, and vice versa. It has been considered as appearance the selection of one or more features from the group. The same is displayed in Table 4 using LSQD.

At the beginning, the system selects groups G_3 , G_4 and G_5 in almost 100% of the cases because of the low threshold imposed (0.5 MaxMFLOPS). When we increase this value to 1 MaxMFLOPS, the spectral features appear. If the restriction is established in 1.5 MaxMFLOPS, the MFCCs start to be selected. When we reach higher values of MFLOPS (3.5), group G_2 is selected, which is composed of features related to the pitch. The case of 4.0 MaxMFLOPS allows the algorithm to select whatever it needs, because the sum of all the costs is lower than this value.

In general LSQD works better than LSLD, since the error rate is lower in most cases, specially when the cost constraint is very limiting. The importance of some features is reflected in the table. For instance, when group G_1 -MFCCs and Δ MFCCs- appears (from 1.5 MaxMFLOPS onwards) its appearance is 100%. In fact, the parameter that best reflects the importance of G_1 is the error rate, since it falls significantly when that group appears (in the case of LSLD, from 57.5% of error to 28.5%, and in the case of LSQD, from 41.9% to 23.4%). Something similar happens when G_2 -pitch, HNR and RUF- appears (from 3.5 MaxMFLOPS onwards). Again, its selection rate is 100% and its contribution to the performance of the system is really significant (error falls from 30.1% to 15.7% with LSLD and from 23.8% to 15.5% with LSQD). The importance of pitch could be directly related to the particular frequency that drones present, which is dependent on the size of the device, the number of blades and the speed.

With regard to the rest of features, G_3 seems to work well only when using LSLD because of its high selection rate. The same applies to G_8 , but when using LSQD. Other features seem to be more robust to changes in the detector used (G_5 , G_6 and G_7), since they present high selection rate for both detectors.

3.2 Analysis of the Model of Drone and Other No-drone Sounds

Then, the error obtained in each of the models included in the drone database will be analyzed. Table 5 shows the different models of drone, the duration of each of them and the error obtained. In these results the best constraint and detector in terms of error have been selected from the previous cases (13.4% of error with 4.0 MFLOPS and LSLD).

From Table 5 it can be seen that Parrot AR is the best detected model (0% of error rate), while the worst one is the UDI 817 (50% of error). This could be because of its minor presence in the database. As it can be observed, a large proportion of the database belongs to DJI Phantom 3, which gets an error rate of 12.2%.

As mentioned previously, the dataset was developed including no-drone sounds present in smart city environments, which can be easily confused with the sound of a drone. In Table 6 the no-drone sounds, the duration of them and the error obtained are detailed.

From the results it can be observed that the most confusing sounds are the fire siren, radial saw and construction work (with error rates of 40.7%, 36.4% and 22.5%, respectively). This could be because the fundamental frequency of these sounds is in the range of the drone frequency (one or two hundreds of Hz). Likewise, other sounds like helicopter, excavator, motorbike or plane are really well detected as no-drone sounds, with error rates below 3%. This is especially interesting in the case of other aerial vehicles (helicopter, plane), since they could be more conflicting with drones as they share the same space of work (the sky) and they could appear at the same time.

4 CONCLUSIONS

The aim of this work is to develop a system capable of detecting the presence of drones in real time. To this end, different experiments related to Smart Sound Processing (SSP) have been carried out, including feature extraction, feature selection and detectors. The objective of the algorithms is to minimize the error rate while controlling the computational cost. This has been reached through a constraint in the number of operations per second (MFLOPS).

Related to the features selected, the results show that MFCCs and features related to pitch are the best subsets of features for the problem at hand, for both linear and quadratic detectors. Depending on the desired final error rate and on the resources of the processing device, a compromise should be reached be-

Table 3: Cost, error rate and probability of appearance of the features groups with LSLD.

MaxMFLOPS (MFLOPS)		0.5	1.0	1.5	2.0	2.5	3.0	3.5	4.0
Error Rate (%)		52.3	57.5	28.5	30.4	31.9	30.1	15.7	13.4
Selection Rate (%)	G_1 (MFCC+ Δ MFCC)	0.0	0.0	100.0	100.0	100.0	100.0	100.0	100.0
	G_2 (Pitch+HNR+RUF)	0.0	0.0	0.0	0.0	0.0	0.0	100.0	100.0
	G_3 (STE)	73.9	80.8	89.1	93.7	100.0	100.0	25.7	100.0
	G_4 (EE)	100.0	100.0	100.0	100.0	100.0	100.0	0.0	100.0
	G_5 (ZCR)	91.7	13.6	84.7	83.2	89.0	91.6	0.0	95.3
	G_6 (SR)	0.0	100.0	100.0	100.0	100.0	100.0	74.3	93.9
	G_7 (SC)	0.0	92.9	96.1	100.0	100.0	91.3	74.3	100.0
	G_8 (SF)	0.0	70.2	35.6	40.9	50.6	53.0	15.8	22.9

Table 4: Cost, error rate and probability of appearance of the features groups with LSQD.

MaxMFLOPS (MFLOPS)		0.5	1.0	1.5	2.0	2.5	3.0	3.5	4.0
Error Rate (%)		37.8	41.9	23.4	24.2	22.0	23.8	15.5	15.2
Selection Rate (%)	G_1 (MFCC+ Δ MFCC)	0.0	0.0	100.0	100.0	100.0	100.0	100.0	100.0
	G_2 (Pitch+HNR+RUF)	0.0	0.0	0.0	0.0	0.0	0.0	100.0	100.0
	G_3 (STE)	18.7	11.0	0.0	0.0	0.0	0.0	0.0	0.0
	G_4 (EE)	78.4	46.1	100.0	100.0	100.0	100.0	0.0	100.0
	G_5 (ZCR)	100.0	100.0	95.9	96.4	95.9	100.0	0.0	88.6
	G_6 (SR)	0.0	100.0	100.0	100.0	100.0	100.0	100.0	100.0
	G_7 (SC)	0.0	92.9	74.0	69.9	61.4	67.1	95.4	91.3
	G_8 (SF)	0.0	100.0	100.0	100.0	100.0	100.0	61.2	96.6

Table 5: Error Rate of the different models of drones included in the database.

Model of drone	Duration (s)	Error Rate (%)
DJI Phantom 3	1573	12.2
Cheerson CX10	284	13.0
Eachine Racer 250	171	21.6
Parrot AR	103	0.0
UDI 817	17	50.0

Table 6: Error Rate of the no-drone sound included in the database.

No-drone sound	Duration (s)	Error Rate (%)
Plane	128	3.1
Helicopter	124	0.0
Hair clipper	249	14.1
Construction work	316	22.5
Excavator	147	0.0
Motorbike	150	1.3
Mower	268	8.2
Radial saw	22	36.4
Fire siren	135	40.7
Drag racer	55	7.1

tween the two parameters. On the one hand, if the system requires high performance (13.4% of error rate), the solution should include both the MFCCs and the features related to pitch, with at least 3.5 MFLOPS. On the other hand, a worst solution in terms of error rate could be reached (23.4%), but only using 1.5 MFLOPS in the system. Regarding to the detectors,

the results are better in quadratic case, specially when the cost constraint is very restrictive.

In conclusion, the experiments developed show that it is feasible to implement a real time system capable of detecting drone presence in an autonomous way. That is possible thanks to the low cost features proposed in the manuscript, which can be supported by nowadays microprocessors.

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