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A Comparative Study of Machine Learning Methods for Avian Speaker Recognition
          Abstract
          We perform speaker recognition on the calls of Thrush Nightingales (Luscinia luscinia) with a variety of machine learning algorithms. This involves data
          preparation and call extraction, feature generation and the use of standard libraries for supervised classification. The features used are the Mel-frequency
          Cepstral Coefficients (MFCCs), a widely employed feature in speech domains. We analyze our model by considering robustness to training data selection and
          size. Future directions and domain specific challenges and sensitivies are considered.
          Contents
          1. Data Exploration
          2. Feature Generation
          3. Classification and Optimisation
          4. Visualisations
         1. Data Exploration
          In this section we import the data and give an initial exploration via some visualisations.
          Example
 In [1]: import sklearn
          import numpy as np
          import pandas as pd
          import scipy.io.wavfile as sp
          import matplotlib.pyplot as plt
          from os import listdir
          import python_speech_features as psf
          Plot of example signal.
 In [2]: raw_signal = sp.read("./data/XC247266.wav")
          signal = (raw_signal[1][:, 0] + raw_signal[1][:, 1]) / 2
 In [3]: plt.figure(0, figsize = (10, 5))
          time = np.linspace(0, len(signal), num=len(signal)) / raw_signal[0]
          plt.title('Example Signal')
          plt.plot(time, signal)
          plt.show()
                                              Example Signal
            15000
            10000
             5000
            -5000
           -10000
           -15000
                                          20
                   Ó
          Call Detection
          As illustrated by the above figure, we must also detect calls to identify the relevent portions of the time series.
 In [4]: def call_detector(signal, threshold, dispersion, rate, window):
              separation = dispersion * rate
              noise = separation + 1
              output = []
              for i in xrange(len(signal) - window - 1):
                  if np.mean(signal[i : i + window]) < threshold:</pre>
                      noise += 1
                  elif np.mean(signal[i : i + window]) > threshold:
                      if noise > separation:
                           output.append(i)
                      noise = 0
              return output
          The above algorithm is used to compute call onset and offset times.
 In [5]: call_onsets = np.asarray(call_detector(signal, 200, 2, 44100, 30))
          call_offsets = np.asarray(call_detector(signal[::-1], 200, 2, 44100, 30))
          for i in xrange(len(call_offsets)):
              call_offsets[i] = len(signal) - call_offsets[i] - 1
          call_offsets = np.asarray(call_offsets[::-1])
          Example output from the call detection process.
 In [6]: [[call_onsets[i], call_offsets[i]] for i in xrange(min(len(call_onsets), len(call_offsets)))]
 Out[6]: [[64772, 208065],
           [356571, 553274],
           [683573, 1110755],
           [1205287, 1278861],
           [1401718, 1581712],
           [1728687, 1921742],
           [2072717, 2256330]]
 In [7]: plt.figure(0, figsize=(10, 5))
          time = np.linspace(0, len(signal), num=len(signal)) / raw_signal[0]
          plt.title('Example Automatic Call Detection')
          plt.plot(time, signal)
          for onset in call_onsets:
              plt.axvline(x = onset / 44100, color = 'g')
          for offset in call_offsets:
              plt.axvline(x = offset / 44100, color = 'r')
          plt.show()
                                       Example Automatic Call Detection
            15000
            10000
             5000
           -10000
           -15000
          Read audio and construct dictionary to store numeric representation
 In [8]: audio_files = listdir("./data")
          rate = sp.read("./data/" + audio_files[1])[0]
          ids = [x[0:7] for x in listdir("./data")]
          signals = dict.fromkeys(ids)
          for i, f in enumerate(audio_files):
              raw_signal = sp.read("./data/" + f)
              signals[ids[i]] = (raw_signal[1][:, 0] + raw_signal[1][:, 1]) / 2
          signals
 Out[8]: {'XC24726': array([ 0, -1, -1, ..., -16, -38, -34], dtype=int16),
           'XC24779': array([-26, -26, -7, ..., -41, -63, -51], dtype=int16),
           'XC29044': array([ 51, 71, 61, ..., -60, -46, -13], dtype=int16),
           'XC30043': array([0, 0, 0, ..., 0, 0, 0], dtype=int16),
           'XC31658': array([ -1, -1, 1, ..., 8, -80, -111], dtype=int16),
           'XC31879': array([ 0, 0, ..., -24, 61, 108], dtype=int16),
           'XC36993': array([ 0, 0, 0, ..., -2, -2, -1], dtype=int16),
           'XC37067': array([ 0, 0, ..., -3882, -4112, -4262], dtype=int16),
           'XC37080': array([ 24, -6, -47, ..., 0, 0, 0], dtype=int16)}
 In [9]: def extract_calls(signal, threshold, dispersion, rate, window):
              # detect call starts in forward and back direction
              call_onsets = np.asarray(call_detector(signal, threshold, dispersion, rate, window))
              call_offsets = np.asarray(call_detector(signal[::-1], threshold, dispersion, rate, window))
              # align indices and order of call offsets
              for i in xrange(len(call_offsets)):
                  call_offsets[i] = len(signal) - call_offsets[i] - 1
              call_offsets = np.asarray(call_offsets[::-1])
              # combine to form a list of lists containing call onset and offset times
              return [[call_onsets[i], call_offsets[i]] for i in xrange(min(len(call_onsets), len(call_offsets)))]
          The following collects the call offset and onset times for each recording/identity.
In [10]: calls = dict.fromkeys(ids)
          for identity in ids:
              calls[identity] = extract_calls(signals[identity], 200, 2, rate, 30)
          Next, we remove the portions of noise or silence from the signal.
In [11]: filt_signals = dict.fromkeys(ids)
          for i in xrange(len(ids)):
              filt_signals[ids[i]] = np.concatenate([signals[ids[i]][x[0]:x[1]]  for x in calls[ids[i]]])
          filt_signals
Out[11]: {'XC24726': array([-655, -156, 405, ..., 949, 667, 306], dtype=int16),
           'XC24779': array([ 437, 1923, 2374, ..., 855, 719, 528], dtype=int16),
           'XC29044': array([1696, 1942, 1784, ..., -961, 1308, 2552], dtype=int16),
           'XC29625': array([-2581, -440, 1861, ..., 882, 816, 674], dtype=int16),
           'XC30043': array([ 81, 330, 536, ..., 1098, 934, 652], dtype=int16),
           'XC31658': array([-1445, -963, -343, ..., 595, 293, -38], dtype=int16),
           'XC31879': array([ 297, 228, 175, ..., -165, -308, -362], dtype=int16),
           'XC36993': array([ 87, 5, -87, ..., 1047, 1122, 1005], dtype=int16),
           'XC37067': array([15675, 14476, 14413, ..., -248, -268, -279], dtype=int16),
           'XC37080': array([-647, 466, 1492, ..., 1891, 1913, 1376], dtype=int16)}
          2. Feature Generation
          In this section we transform the above signal time series data into suitable features for use in classification. In particular, we generate the Mel-frequency
          Cepstral Coefficients (MFCCs), a commonly used feature in speech classification. This is an important step, as a large time series passed to a machine
          learning algorithm would have prohibitively large dimension - as per the curse of dimensionality. However, the MFCC feature vector reduces this to have only
          13 dimensions.
In [12]: from sklearn.model_selection import train_test_split
In [13]: features = np.zeros(14)
          for i in xrange(len(ids)):
              mf = psf.mfcc(filt_signals[ids[i]], samplerate = rate, winlen=0.025, winstep=0.01, numcep=13,
                            nfilt=26, nfft=512, lowfreq=0, highfreq = None, preemph=0.97,
                            ceplifter=22, appendEnergy=True)
              a = np.zeros((np.shape(mf)[0], 14))
              a[:, : -1] = mf
              a[:, -1] = np.repeat(i, np.shape(mf)[0])
              features = np.vstack((features, a))
          features = features[1:, ]
In [15]: X_train, X_test, y_train, y_test = train_test_split(features[:, 0:12], features[:, 13],
                                                                test_size=0.33, random_state=42)
          3. Classification and Optimisation
In [17]: from sklearn.tree import DecisionTreeClassifier
          from sklearn.ensemble import RandomForestClassifier
          from sklearn import svm
          from sklearn.ensemble import AdaBoostClassifier
          from sklearn.naive_bayes import GaussianNB
          from sklearn import metrics
          from sklearn.model_selection import GridSearchCV
          Decision Tree
In [16]: dt = DecisionTreeClassifier()
          dt.fit(X_train, y_train)
          dt_acc = dt.score(X_test, y_test)
          dt_f1 = metrics.f1_score(y_test, dt.predict(X_test), average = 'macro')
          print [dt_acc, dt_f1]
          [0.76317379338121771, 0.6957135133088832]
          Exhaustive search parameter optimisation with GridSearchCV.
In [19]: parameters = [
            {'criterion': ['gini', 'entropy'], 'min_samples_split' : [2, 3, 4, 5, 6, 7, 8],
             'splitter' : ['best', 'random']},
          dt_gs = GridSearchCV(DecisionTreeClassifier(), parameters)
          dt_gs.fit(X_train, y_train)
          dt_opt_acc = dt_gs.score(X_test, y_test)
          dt_opt_f1 = metrics.f1_score(y_test, dt_gs.predict(X_test), average = 'macro')
          print [dt_opt_acc, dt_opt_f1]
          [0.76639591864133716, 0.70030897539978509]
          Random Forest
In [21]: rf = RandomForestClassifier()
         rf.fit(X_train, y_train)
         rf_acc = rf.score(X_test, y_test)
          rf_f1 = metrics.f1_score(y_test, rf.predict(X_test), average = 'macro')
          print [rf_acc, rf_f1]
          [0.83808820567899578,\ 0.78196962371116441]
          Exhaustive search parameter optimisation with GridSearchCV.
In [19]: parameters = [
            {'criterion': ['gini', 'entropy'], 'min_samples_split' : [2, 4, 6, 8],
             'n_estimators': [10, 100, 200, 400, 500]}]
          rf_gs = GridSearchCV(RandomForestClassifier(), parameters)
          rf_gs.fit(X_train, y_train)
          rf_opt_acc = rf_gs.score(X_test, y_test)
          rf_opt_f1 = metrics.f1_score(y_test, rf_gs.predict(X_test), average = 'macro')
          print [rf_opt_acc, rf_opt_f1]
          [0.88574880848492987, 0.84150675534700559]
          Support Vector Machine
In [20]: sm = svm.SVC()
          sm_fit = sm.fit(X_train, y_train)
          print sm_fit.score(X_test, y_test)
          0.201651339196
          Exhaustive search parameter optimisation with GridSearchCV.
In [21]: parameters = [\{'C' : [0.5, 1.0, 1.5, 2.0]\}]
          sm_gs = GridSearchCV(svm.SVC(), parameters)
          sm_gs.fit(X_train, y_train)
          sm_opt_acc = sm_gs.score(X_test, y_test)
          print sm_opt_acc
          0.20265825334
          Gaussian Naive Bayes
In [22]: | gnb = GaussianNB()
          gnb.fit(X_train, y_train)
          gnb_acc = gnb.score(X_test, y_test)
          gnb_f1 = metrics.f1_score(y_test, gnb.predict(X_test), average = 'macro')
          print [gnb_acc, gnb_f1]
          [0.6198563469154863, 0.53774934409910324]
          Optimisation not applicable, due to lack of informative priors.
          AdaBoost
In [23]: ab = AdaBoostClassifier()
          ab.fit(X_train, y_train)
          ab_acc = ab.score(X_test, y_test)
          ab_f1 = metrics.f1_score(y_test, ab.predict(X_test), average = 'macro')
          print [ab_acc, ab_f1]
          [0.62858293616164329, 0.52879537935931309]
          Exhaustive search parameter optimisation with GridSearchCV.
In [24]: parameters = [{'n_estimators' : [50, 100, 200, 500], 'learning_rate' : [0.5, 1.0, 1.5, 2.0]}]
          ab_gs = GridSearchCV(AdaBoostClassifier(), parameters)
          ab_gs.fit(X_train, y_train)
          ab_opt_acc = ab_gs.score(X_test, y_test)
          ab_opt_f1 = metrics.f1_score(y_test, ab_gs.predict(X_test), average = 'macro')
          print [ab_opt_acc, ab_opt_f1]
          [0.64127005437336382, 0.49441287605529649]
          5. Visualisations
          The following bar chart visualises the relative feature importances for the fitted Random Forest classifier.
In [32]: plt.figure(0, figsize = (10, 5))
          plt.bar(range(1, 13), rf.feature_importances_)
          plt.show()
           0.16
           0.14
           0.12
           0.10
           0.08
           0.06
           0.04
           0.02
          Next, we visualise performance over successive distinct training and testing splits to test robustness.
In [33]: out = np.zeros(100)
          for i in xrange(100):
              X_train, X_test, y_train, y_test = train_test_split(features[:, 0:12], features[:, 13],
                                                                test_size=0.33, random_state= i)
              rf = RandomForestClassifier()
              rf.fit(X_train, y_train)
              out[i] = rf.score(X_test, y_test)
In [35]: plt.figure(0, figsize = (10, 5))
          plt.plot(range(100), out)
          plt.show()
           0.848
           0.846
           0.844
           0.842
           0.840
           0.838
           0.836
          Finally, we consider how the quantity fo training data effects classification accuracy.
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In [42]: out2 = np.zeros(9)

plt.show()

0.84

0.82

0.80

0.78

for i **in** xrange(10, 100, 10):

rf.fit(X_train, y_train)

plt.plot(range(10, 100, 10), out2)

In [43]: plt.figure(0, figsize = (10, 5))

rf = RandomForestClassifier()

 $out2[i/10 - 1] = rf.score(X_test, y_test)$

X_train, X_test, y_train, y_test = train_test_split(features[:, 0:12], features[:, 13],

test_size = i/100.0, random_state= 42)

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