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Заключение

Список литературы

Приложение. Листинг программы

Список литературы

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Приложение. Листинг программы

main.py

```
# -*- coding: utf-8 -*-
import pylab
import datetime
import sklearn
from sklearn import metrics, ensemble, cluster
import os
import recognition as rec
os.system('cls')
startTime = datetime.datetime.now()
print startTime
CONST setN=150
CONST esimators=100
CONST depth=25
CONST learner='RandomForestClassifier'
CONST supvec='Kmeans-2, mfcc = 26'
#Parsing,getting paths, making list
speakerInfoFile=open('speaker-info.txt')
speakerList=rec.parse(speakerInfoFile)
print 'Parsing done',str(datetime.datetime.now()-startTime)
rightTrainList=rec.makeSublist(speakerList,0,CONST setN)
rightTestList=rec.makeSublist(speakerList,0,10)
print 'RightSpeakerListsdone',str(datetime.datetime.now()-startTime)
directory = 'wav48'
audiofiles=rec.getFiles(directory)
print "Getting files'paths done", str(datetime.datetime.now()-startTime)
trainFiles=rec.createFilesList(audiofiles,0,CONST setN)
testFiles=rec.createFilesList(audiofiles,CONST setN,CONST setN+10)
print 'Files list done',str(datetime.datetime.now()-startTime)
filename='results.txt'
txt = open(filename, 'a')
#Supervectors, RandomForestClassifier
trainSupVec=rec.getMfcc(trainFiles)
```

```
testSupVec=rec.getMfcc(testFiles)
      print 'Make supervectors done',str(datetime.datetime.now()-startTime)
      clf = ensemble.RandomForestClassifier(n estimators=CONST esimators,
                                           max depth=CONST depth)
      clf = clf.fit(trainSupVec, rightTrainList)
      print 'RandomForestClassifier
done',str(datetime.datetime.now()-startTime)
      #Prediction
      predictTest=clf.predict(testSupVec)
      predictTrain=clf.predict(trainSupVec)
      print 'Prediction done', str(datetime.datetime.now()-startTime)
      #Estimation
      resultTest=rec.intersec(predictTest, rightTestList)
      resultTrain=rec.intersec(predictTrain, rightTrainList)
      errlst=[]
      for x in xrange(len(predictTest)):
       if predictTest[x]!=rightTestList[x]:
            errlst.append((rightTestList[x],predictTest[x]))
      lenTrain all,lenTrain res,perTrain=rec.estimate(rightTrainList,
resultTrain)
      lenTest all,lenTest res,perTest=rec.estimate(rightTestList, resultTest)
      #Results output
      now=str(datetime.datetime.now().date())+" "+\
            str(datetime.datetime.now().time())
      cond="set = "+str(CONST setN)+\
          ", trees = "+str(CONST esimators)+\
          ", max_depth = "+str(CONST_depth)
      resTrain="Train set: size = "+str(lenTrain all)+\
         ", rigth = "+str(lenTrain res)+", "+str(perTrain)
      resTest="Test set: size = "+str(lenTest all)+\
             ", rigth = "+str(lenTest res)+", "+str(perTest)
      txt.write(now+"\n"+
            CONST learner+"\n"+
            CONST supvec+"\n"+
```

```
cond+"\n"+
            resTrain+"\n"+
            resTest+"\n\n\n")
      txt.close()
      print 'time elapsed =', (datetime.datetime.now() - startTime)
      print '\a'
      recognition.py
      # -*- coding: utf-8 -*-
      import os
      import sklearn
      from sklearn.cluster import KMeans, MiniBatchKMeans
      from features import mfcc
      from features import logfbank
      import scipy.io.wavfile as wav
      from sklearn import mixture
      import numpy as np
      def parse(inputFile):
            11 11 11
            Parsing input file into dictionary
            :param inputFile: lines like "225 23 F English
                                                                      Southern
England"
            :return outputDict:list of No. of speakers
            temp(dictionary like {id:(age, gender, accents, region)}
            Sometimes there is no region)
            ** ** **
            indexes=[]
            genders=[]
            ages=[]
            accents=[]
            regions=[]
            for inputString in inputFile:
            text=''
            symbol = 0
            #Indexes
            while symbol < 3 and inputString[symbol]!=' ' and</pre>
inputString[symbol]!='\n':
                   text+=inputString[symbol]
                   symbol += 1
            indexes.append(int(text))
             #Ages
            symbol += 2
```

```
text = inputString[symbol]
            symbol += 1
            text +=inputString[symbol]
            ages.append(int(text))
            #Genders
            symbol += 3
            text = inputString[symbol]
            genders.append(text)
            #Accents
            symbol +=5
            text=''
            while inputString[symbol]!=' ' and inputString[symbol]!='\n':
                   text+=inputString[symbol]
                   symbol += 1
            accents.append(text)
            #Region
            text=''
            if(inputString[symbol]!='\n'):
                   while inputString[symbol] == ' ':
                   symbol+=1
                   while (inputString[symbol]!='\n'):
                   text+=inputString[symbol]
                   symbol += 1
            regions.append(text)
            #zipped list = zip(ages, genders, accents, regions)
            #liswit = zip(indexes, zipped list)
            #outputDict = dict(liswit)
            return indexes
      def getFiles(directory):
            Collecting paths to .wav files into one list
             :param directory: path to direcrory
            :return audiofiles: list with paths "wav48/<speaker No>/<speaker
No> <sentence No>
            11 11 11
            audiofiles = []
            lenghs=[]
            for d, dirs, files in os.walk(directory):
            files1=[]
            for onefile in files:
                   files1.append(d+"/"+onefile)
            audiofiles.append(files1)
            audiofiles.pop(0)
            return audiofiles
```

```
def createFilesList(audiofiles, startingFrom, stopAt):
      Divide list into parts
      :param audiofiles: list of paths
      :return trainSet: list of paths
      Sometimes there is no region
      trainSet = []
      for files in audiofiles:
      subSet=[]
      for index in xrange(startingFrom, stopAt):
            subSet.append(files[index])
      trainSet.append(subSet)
      return trainSet
def makeSuperVecAver(path to onefile):
      Making supervector using average values
      :param path to onefile: path to file
      :return aver vec: list of numbers-supervector of mfcc (average)
      11 11 11
      aver vec=[]
      (rate, sig) = wav.read(path to onefile)
      mfcc feat=mfcc(sig, rate, numcep=13)
      aver elem=[0 for x in xrange(len(mfcc feat[0]))]
      for vec in mfcc feat:
      aver elem+=vec
      aver vec=aver elem/len(mfcc feat)
      return aver vec
def makeSuperVecKMean(path to onefile):
      11 11 11
      Making supervector using average values
      :param path to onefile: path to file
      :return aver vec: list of numbers-supervector of mfcc (average)
      clf = KMeans(n clusters=2)
      (rate, sig) = wav.read(path to onefile)
      mfcc feat=mfcc(sig,rate,numcep=26)
      clf.fit(mfcc feat)
      aver vec = listmerge(clf.cluster centers )
      return aver vec
def getMfcc(filesArray):
      Making list of supervector
      :param filesArray: array of paths to files
```

```
:return aver array: list of numbers-supervector of mfcc
(average)
            ** ** **
            aver array=[]
            for vec in filesArray:
            for onefile in vec:
                  aver vec=makeSuperVecKMean(onefile)
                  aver array.append(aver vec)
            return aver array
      def intersec(predict, right):
            result=[]
            for i in xrange(len(predict)):
            if(predict[i] == right[i]):
                  result.append(predict[i])
            return result
      def makeSublist(biglst, startingFrom, stopAt):
            reslst=[]
            for speaker in biglst:
            for i in xrange(startingFrom, stopAt):
                  reslst.append(speaker)
            return reslst
      def estimate(right, res):
            len all=len(right)
            len res=len(res)
            per=(float(len res)/len all)*100
            return len all, len res, per
      def listmerge(lstlst):
            all=[]
            for 1st in 1st1st:
            all.extend(lst)
            return all
```