



WOMEN IN DATA SCIENCE
MYSURU

ML implementations on VoIP



ALTANAI BISHT

Sep 26th

hello, I am **Altanai**

10 yrs in Telecom + Voice Over IP + Media streaming +
Communication as a Service

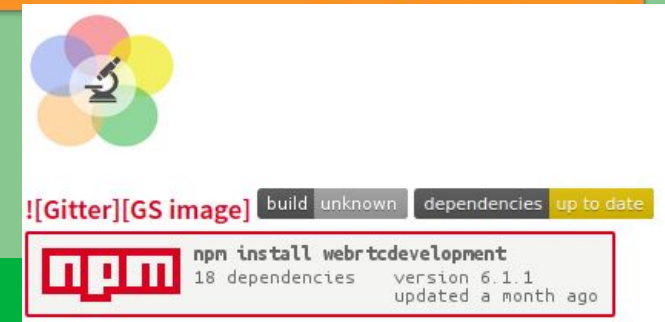
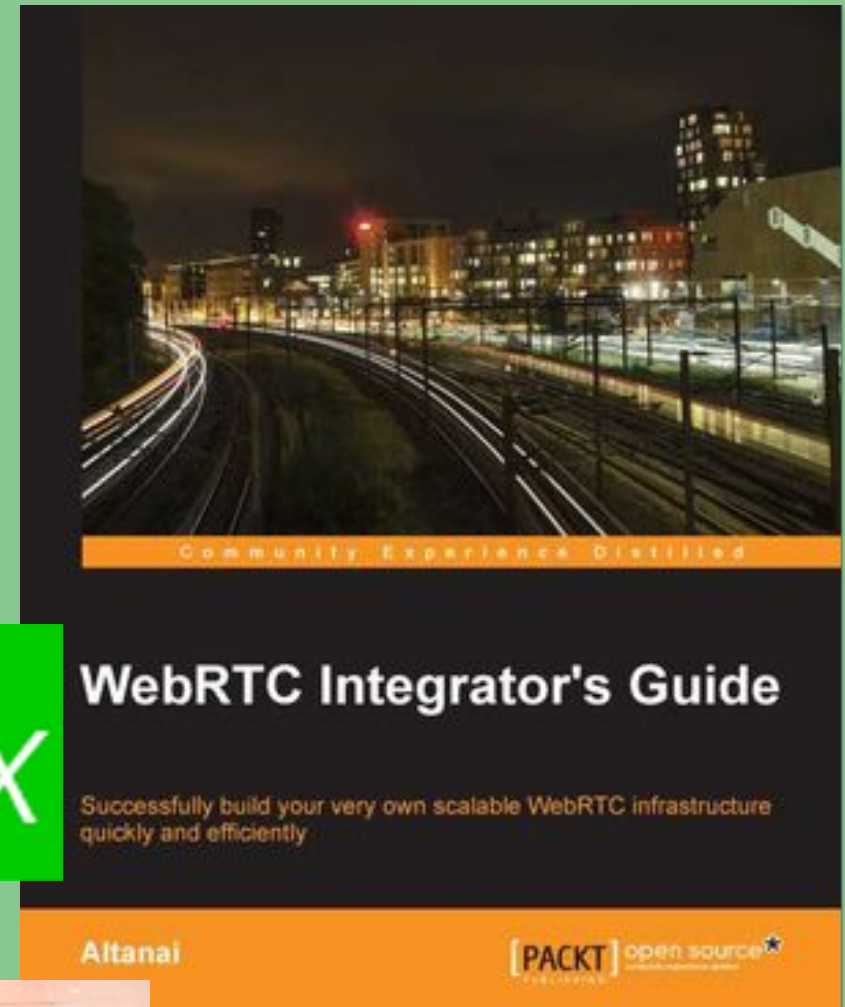
FreeLancer , Open source contributor and blogger

<https://telecom.altanai.com/>

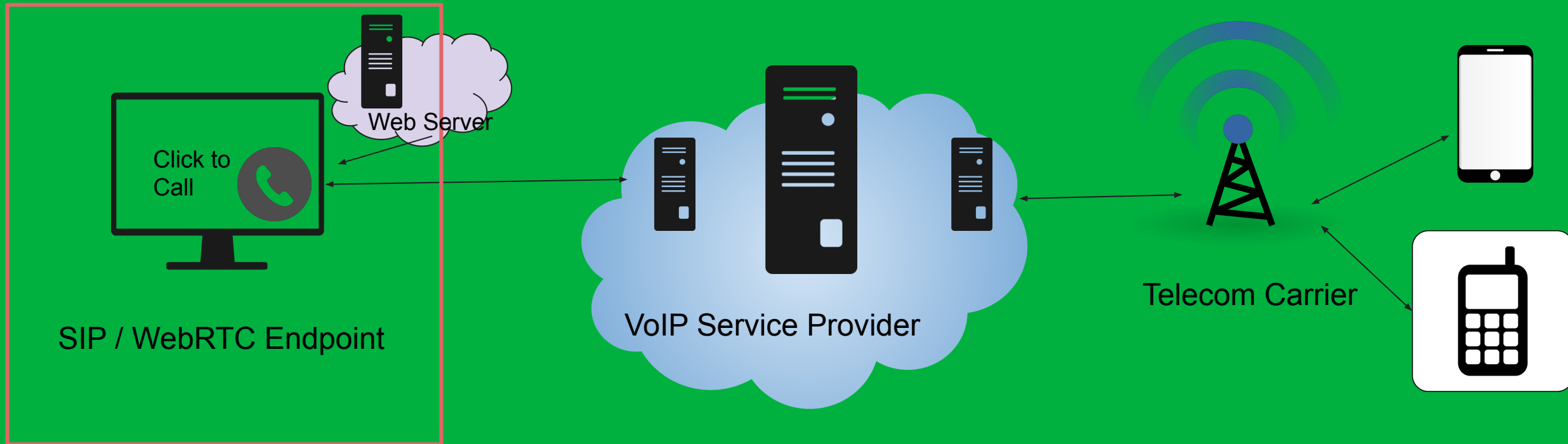
Author of WebRTC Integrator's Guide

Inventor of RamuDroid (Bot to clean roads and
outdoors)

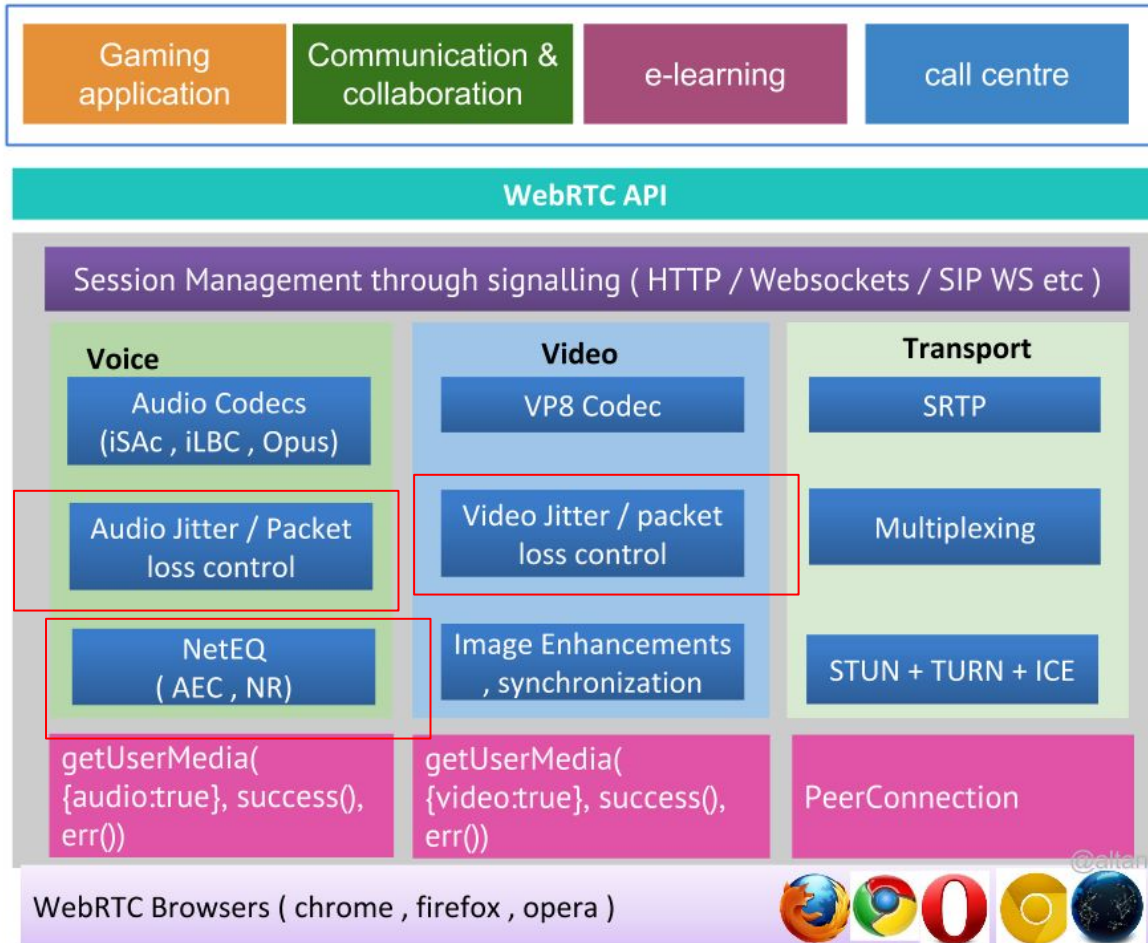
currently with @Airtel



Usual Scenarios ..



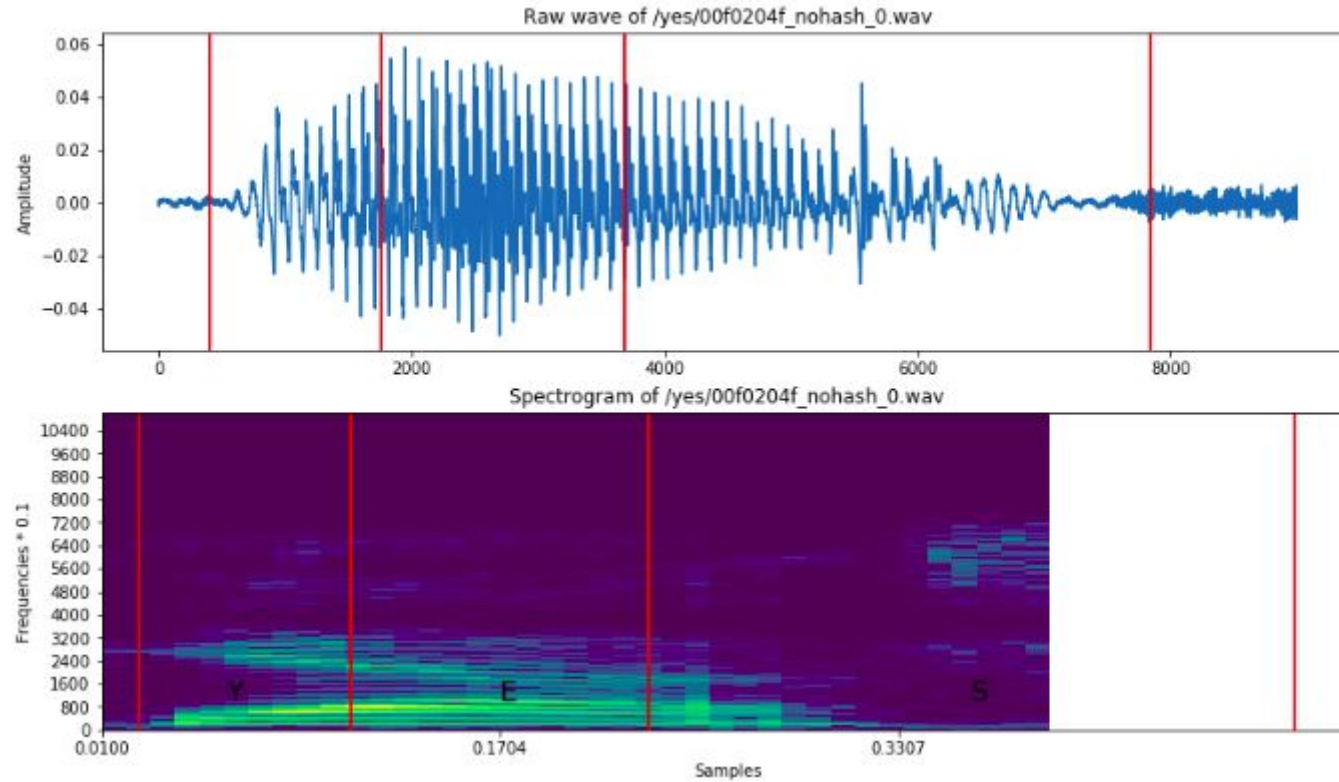
Machine Learning in Media Streams



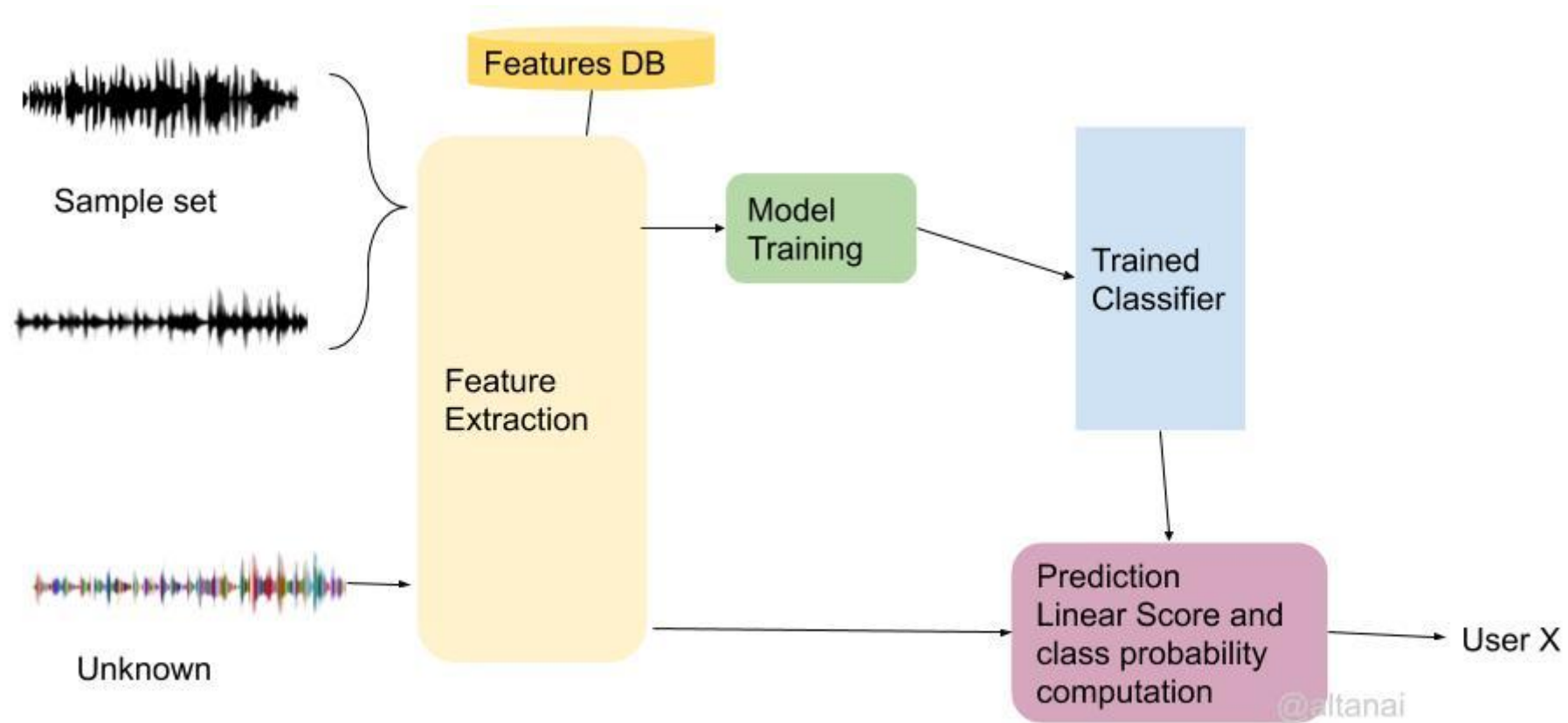
Echo Cancellation
Noise Suppression
Jitter Control
Image Stabilization



Voice Activity Detection



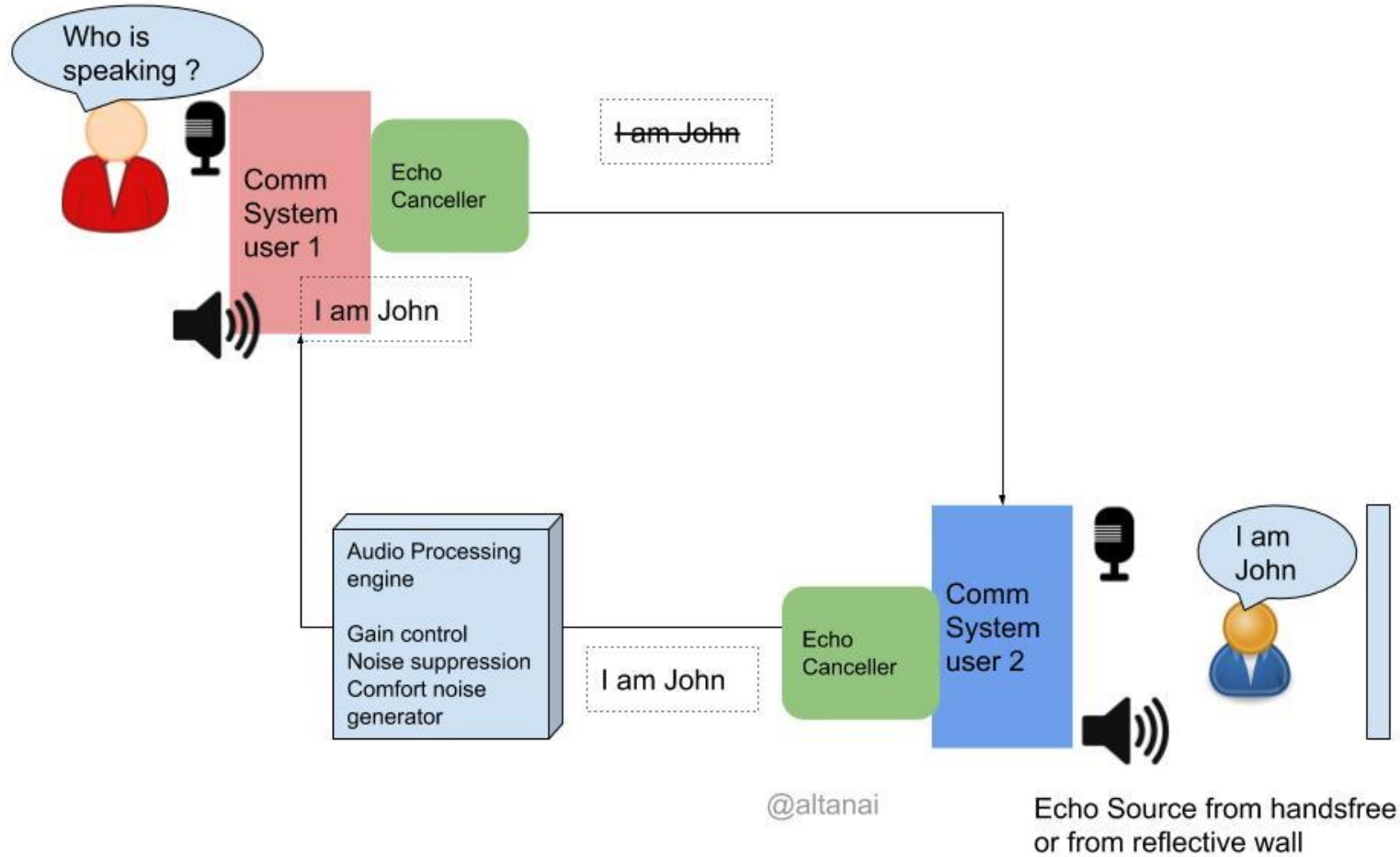
Audio fingerprinting



Supervised learning approach



Echo Cancellation



End device features

Face / Feature Detection

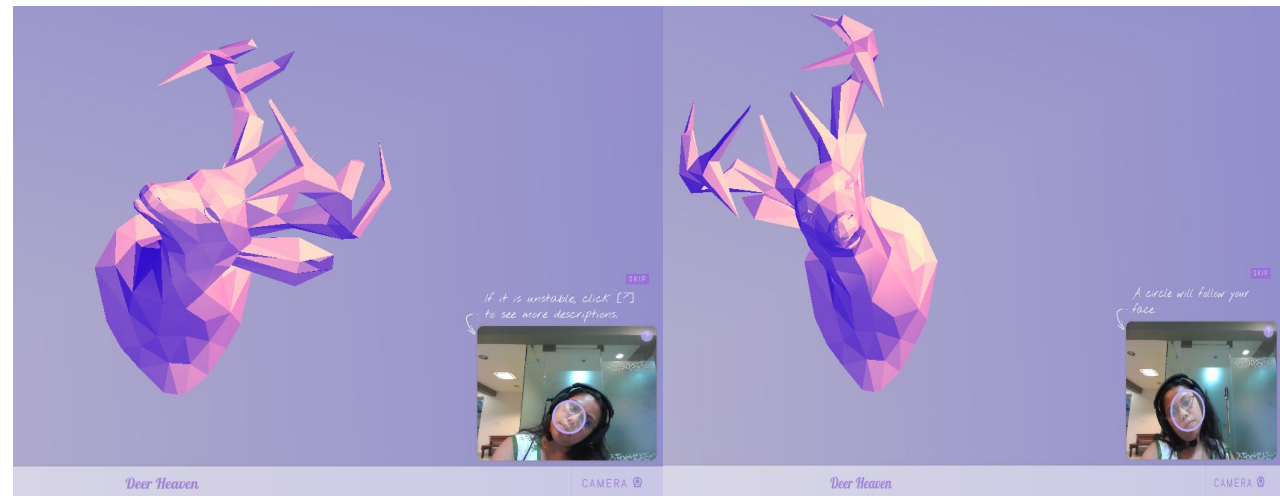
Voice Modulation

Voice Bots / NLP

Background Substitution

Object Recognition

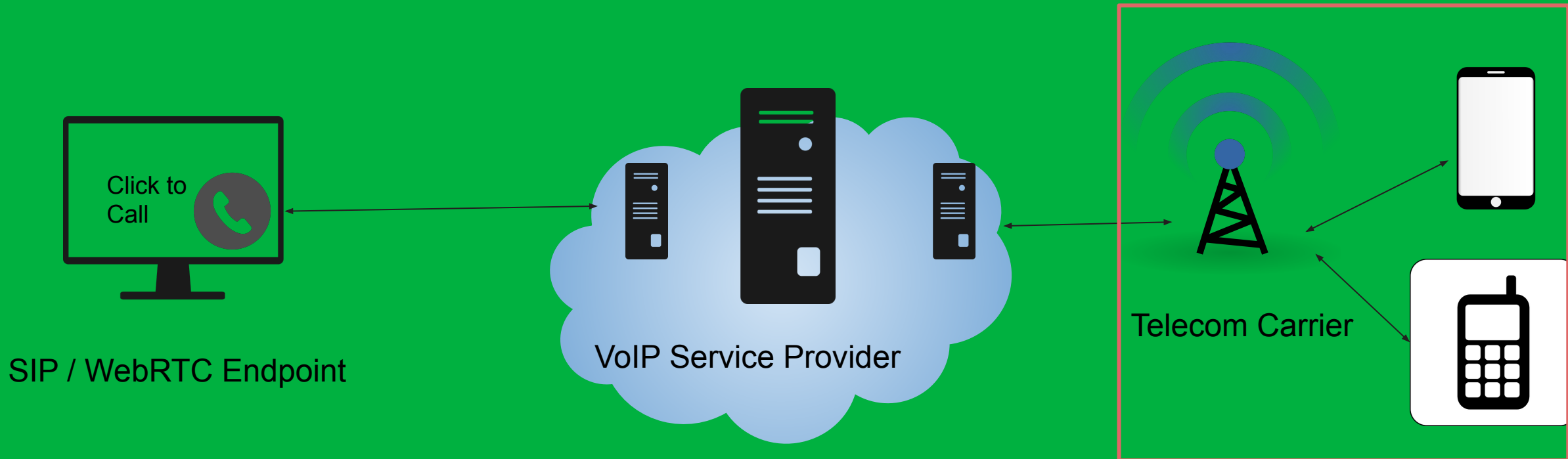
Object Tracking



<http://nayjeejung.com/deerneaven/>



Telecom Service based Applications of ANN

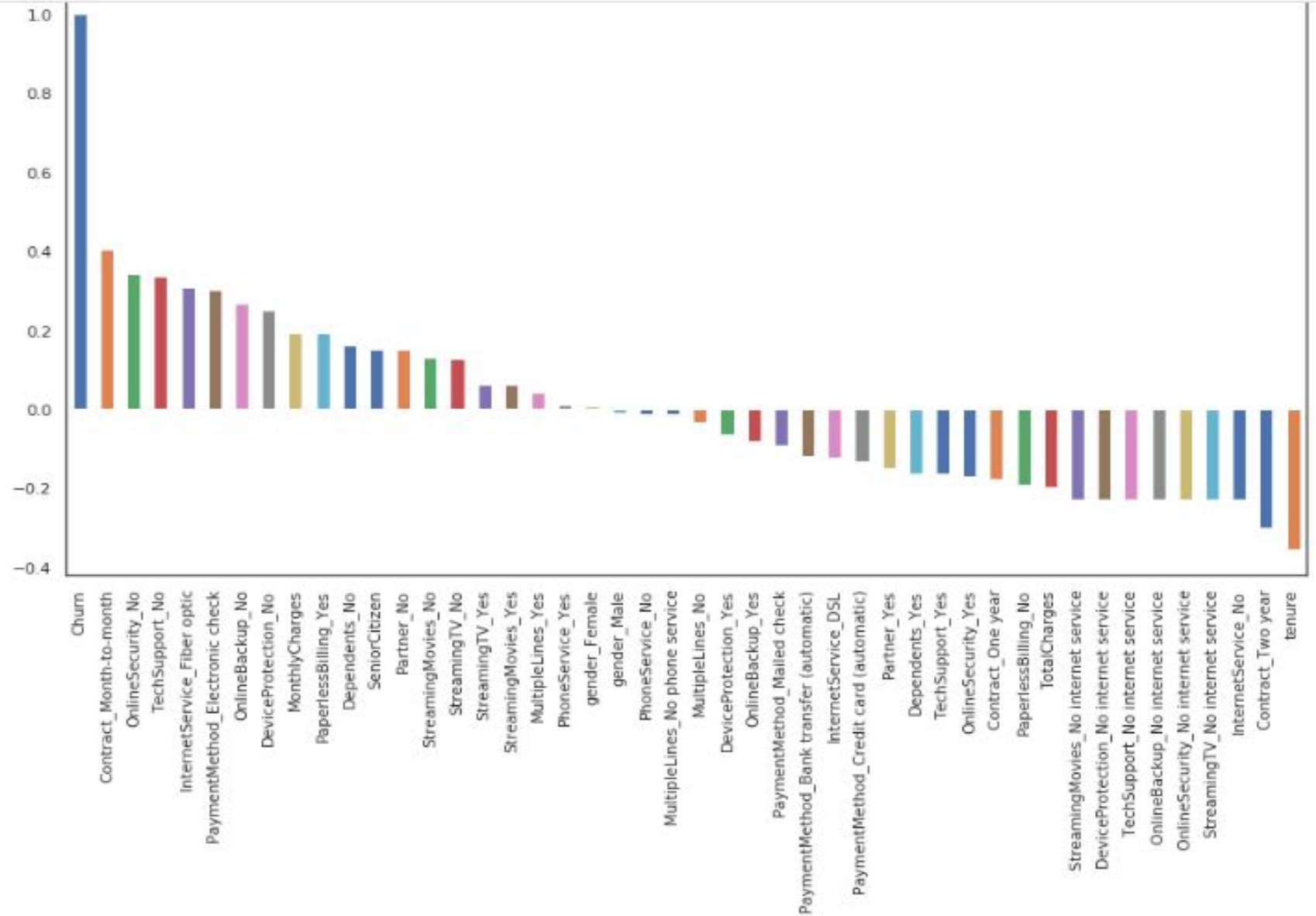


Subscriber Churn and Outliers

Complains ,
Recharges plans

Collect CDR for daily call
patterns

- identify high call volumes, or extremely long calls, or high call volumes from a particular extension



Predictive Analysis

Mean Opinion Score (MOS) - key metric for Quality of Service (QoS) of Call

predicting conversational voice quality non intrusively

Voice quality prediction models and their application in VoIP networks

Publisher: IEEE

[Cite This](#)

[PDF](#)

[Lingfen Sun ; E.C. Ifeachor](#) [All Authors](#)

Language Impact on Voice Quality assessment

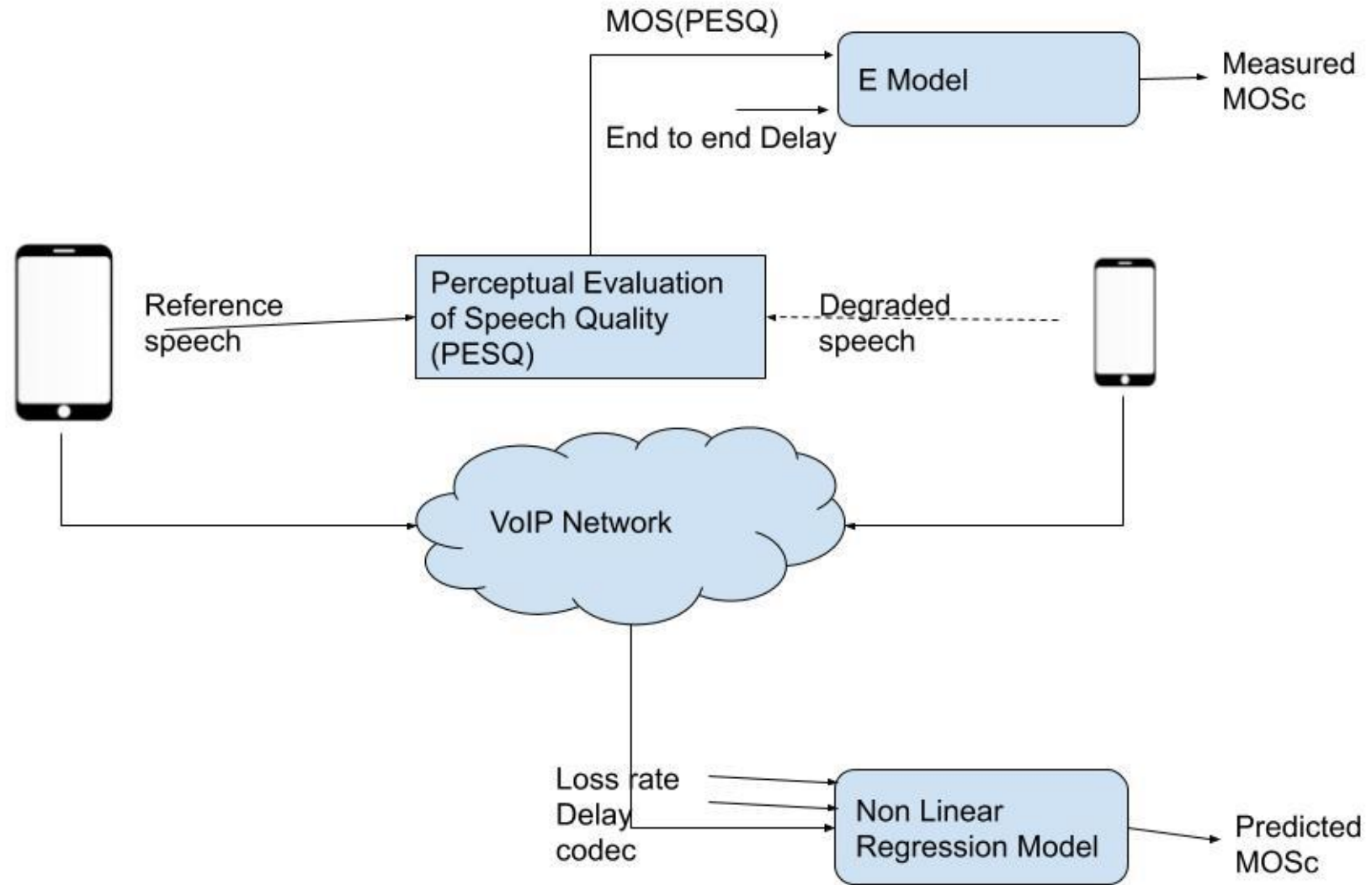
VoIP quality measurement: subjective VoIP quality estimation model for G.711 and G.729 based on native Thai users



Authors: [Therdpong Daengsi](#), [Nalakkhana Khitmoh](#), [Pongpisit Wuttidittachotti](#) [Authors Info & Affiliations](#)

Publication: Multimedia Systems • October 2016 • <https://doi.org/10.1007/s00530-015-0468-3>





Deduce jitter buffer control and adaptive sender bit rate

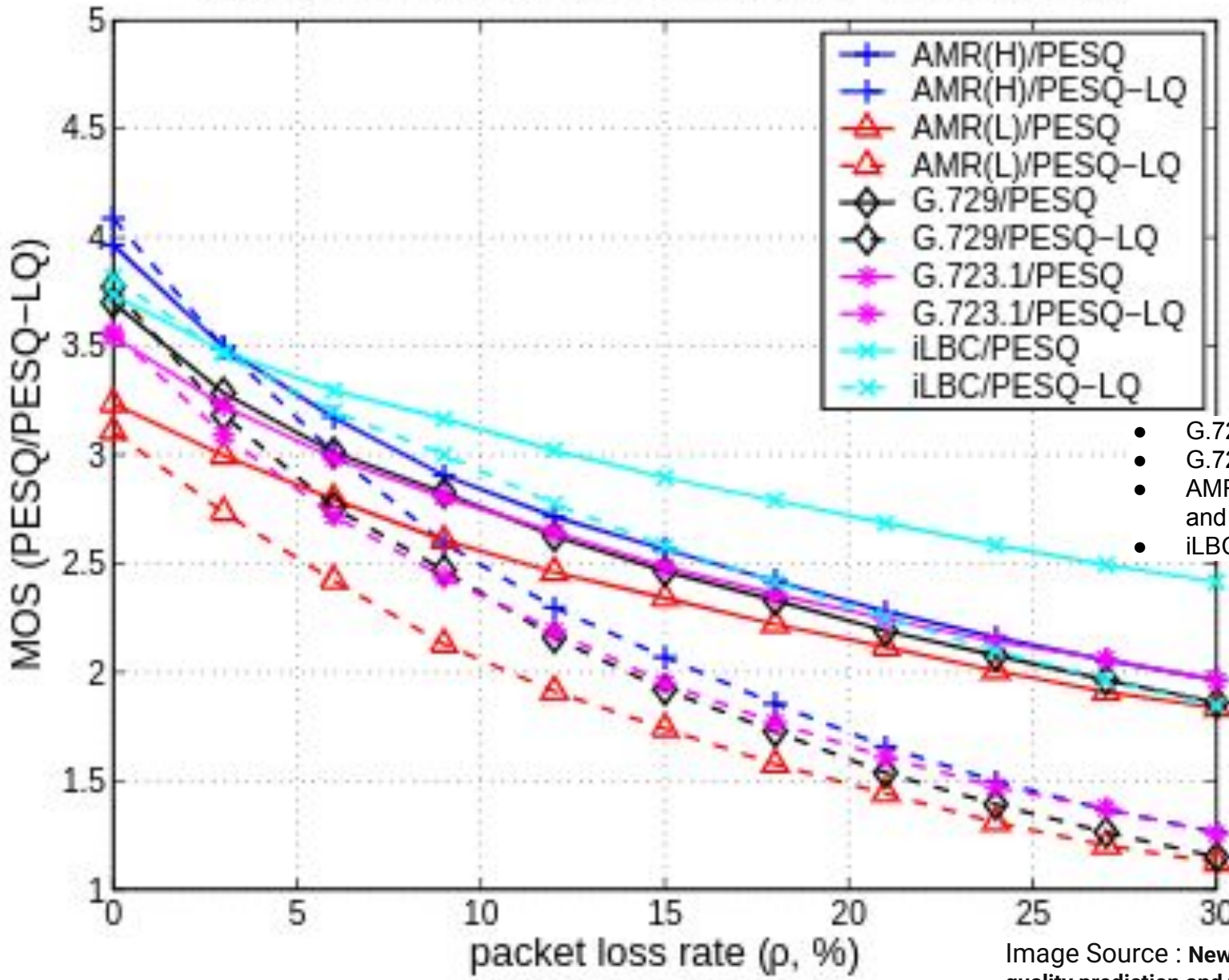
Measurement of conversational voice quality using a combined PESQ and E-model.

Ref : Voice Quality Prediction Models and Their Application in VoIP Networks
Lingfen Sun and Emmanuel C. Ifeachor



Performance Metrics of Packet Loss on Different Codecs

MOS (PESQ/PESQ-LQ) vs. Packet loss rate ρ



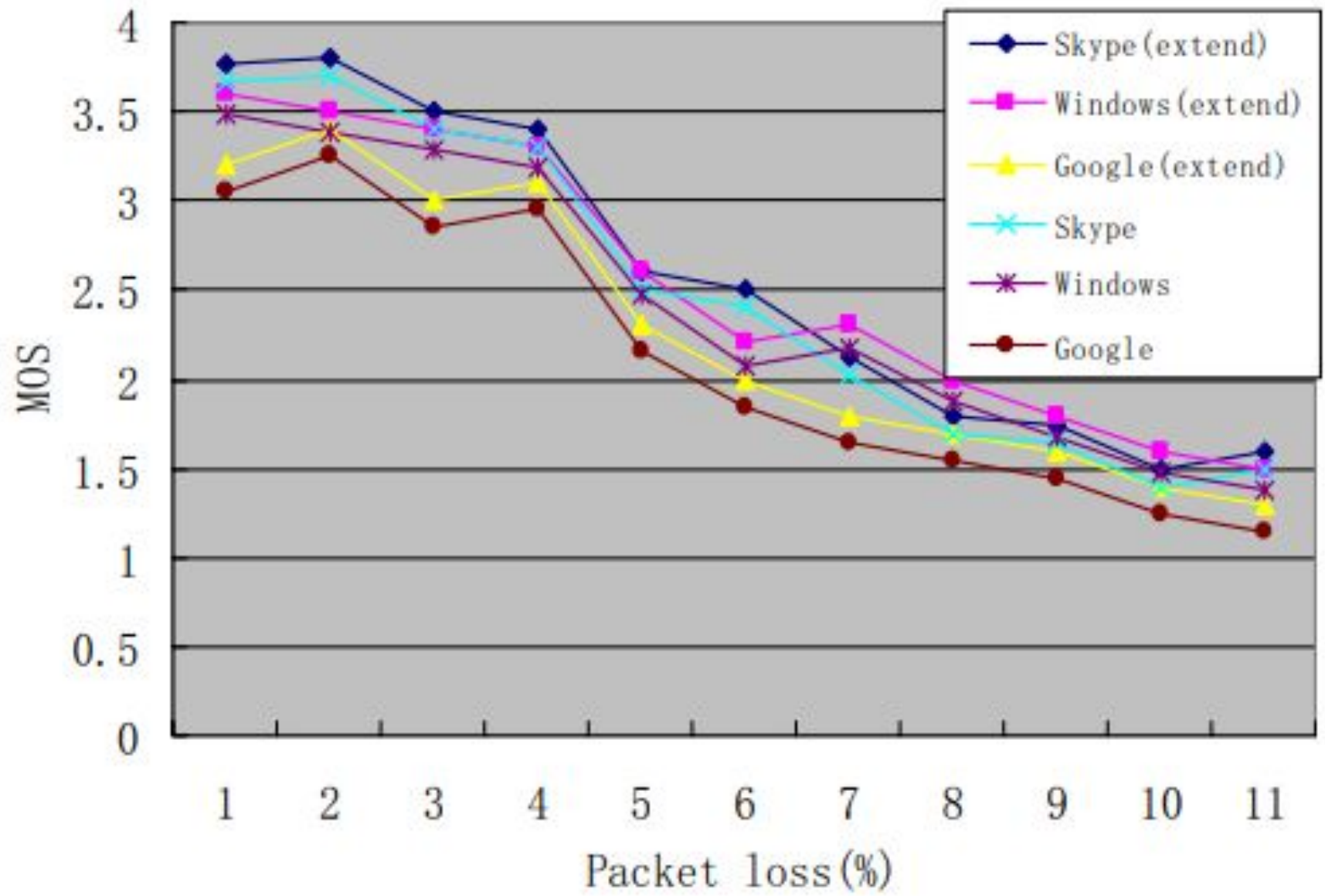
- G.729 (8 Kb/s),
- G.723.1 (6.3 Kb/s),
- AMR(the highest mode, 12.2 Kb/s and the lowest, 4.75 Kb/s)
- iLBC (15.2 Kb/s).

Image Source : New models for perceived voice quality prediction and their applications in playout buffer optimization for VoIP networks
Lingfen Sun , Emmanuel Ifeachor

VoIP provider based Applications of ANN



Performance Metrics of Packet Loss on Different VoIP Service Providers



Source - QoS Evaluation Based on Extend E-Model in VoIP Hongli Zhangab, Zhimin Gua, Zhenqing Tian- 2011

Anomaly detection

- Intrusion detection based on Recurrent Neural Network (RNN) model
- Malicious System Call Sequence Detection (MSCSD)

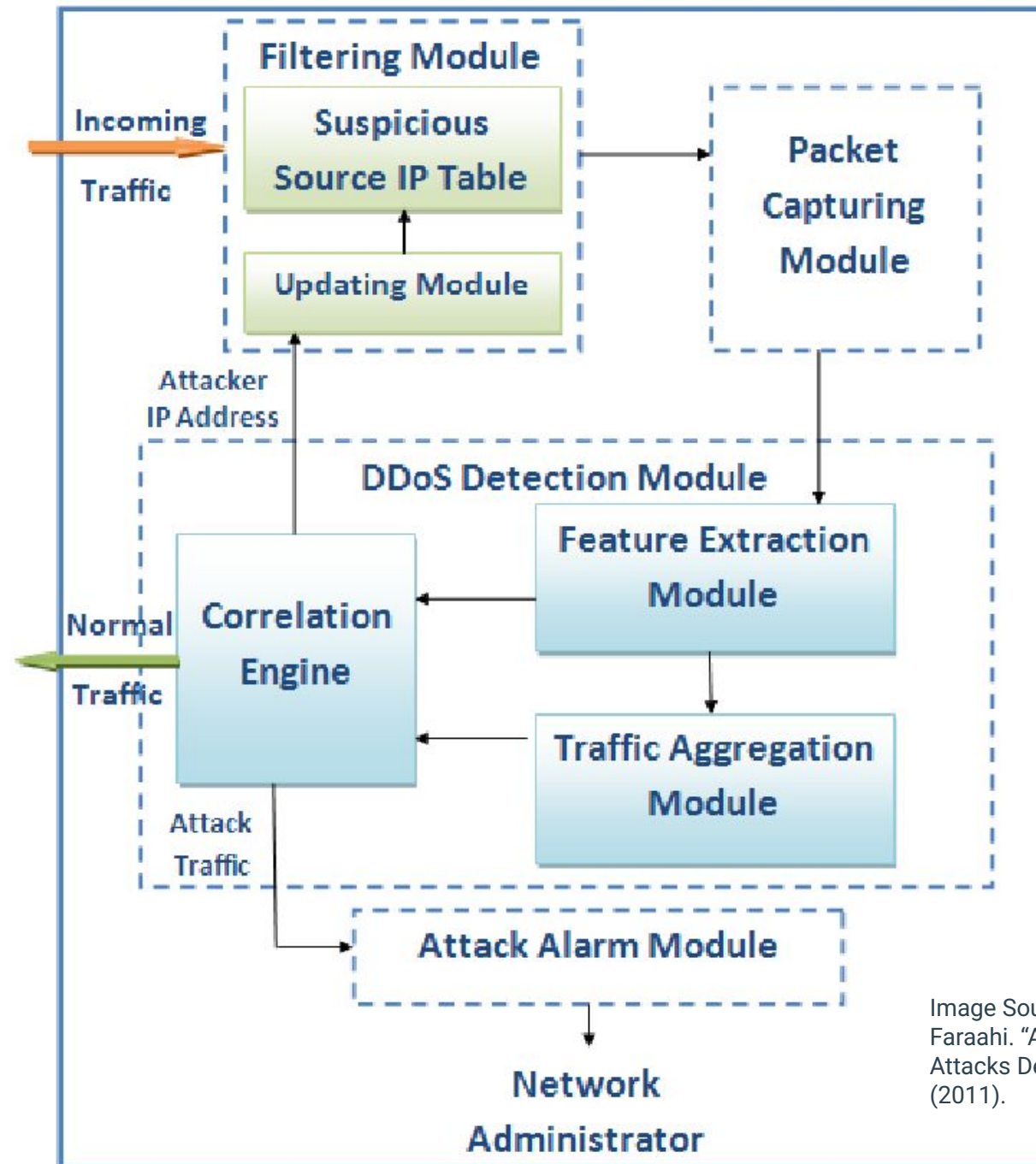


Image Source : Karimzad, Reyhaneh and A. Faraahi. "An Anomaly-Based Method for DDoS Attacks Detection using RBF Neural Networks." (2011).



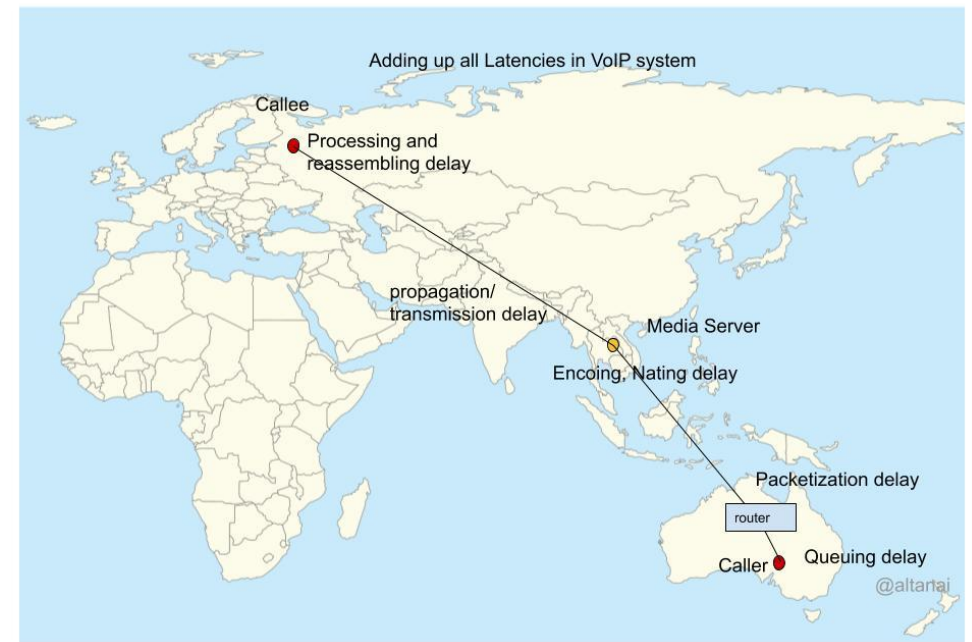
Additional applications of ANN in voIP

Call Prioritization

Geographical routing

Call pattern mapping

- Bypass additional checks to remove latency



Unexplored Areas



Etiquette analysis

Politeness , Courtesy and sticking to matter

- applicable for use cases - Sales , Pre-Sales , Customer Care , Call Centres
- analogous to Sentiment analysis of customers in call

Regulatory analysis

no pin , passwords to be queried from users on phone

- prevent phishing
- for any pincodes use IVR/ DTMF



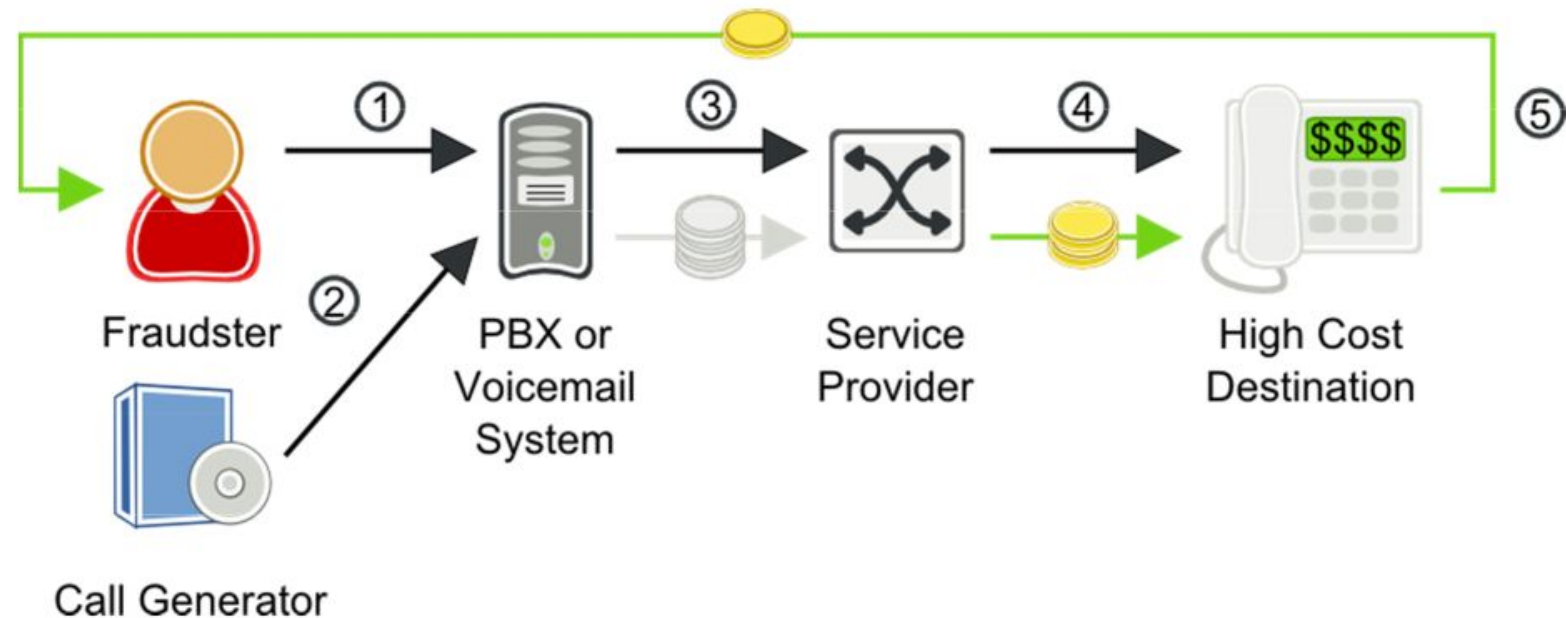
Telecom Fraud

Traffic Pumping

- “access stimulation” techniques to boost traffic to a high cost destination

Defraud Telecom Service Providers

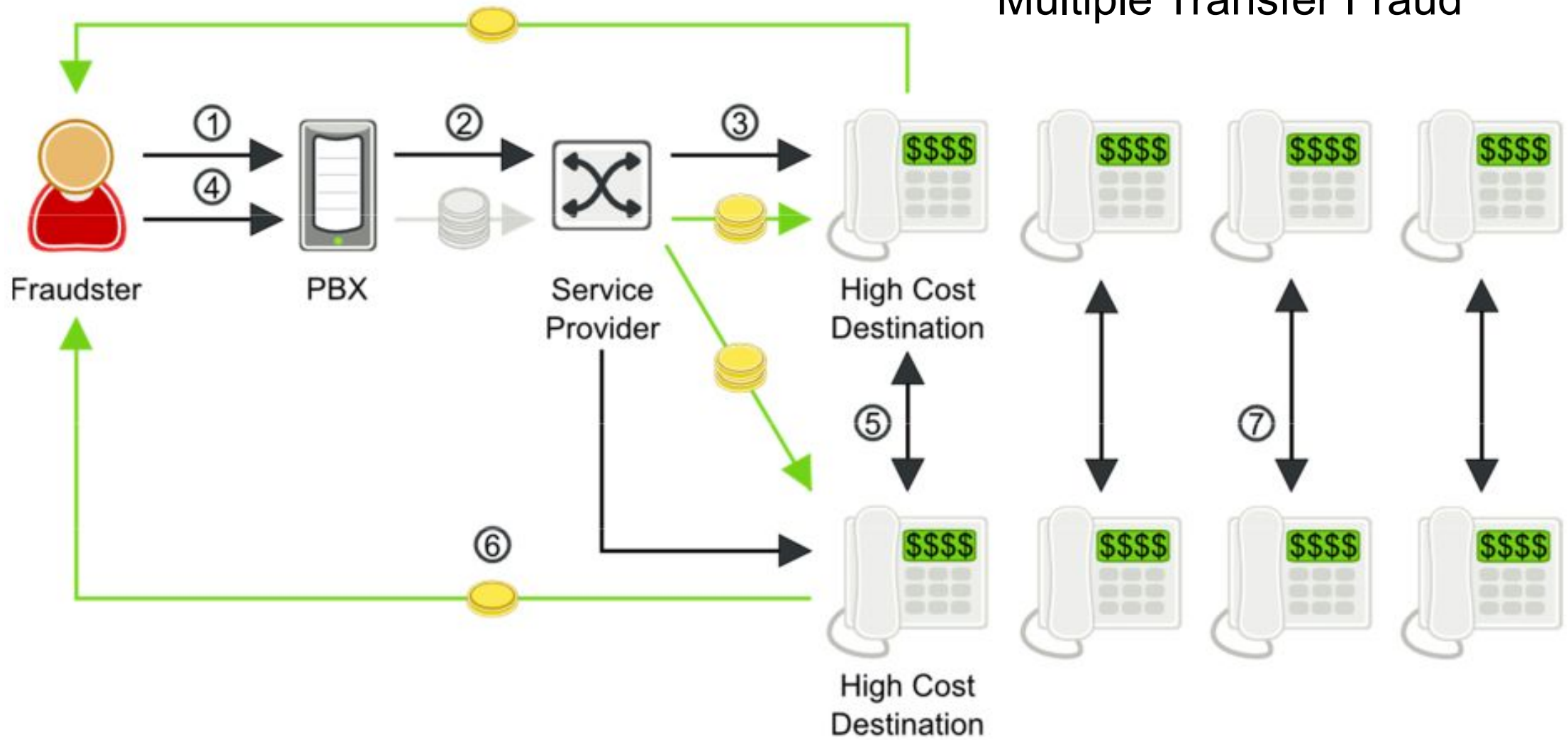
- Exploitation of SIP trunks
- regulatory loopholes
- Premium rate numbers misused



Img source : TransNexus



Multiple Transfer Fraud



Img source : TransNexus

Other popular VoIP vulnerabilities

One ring and Cut to generate Call back revenue

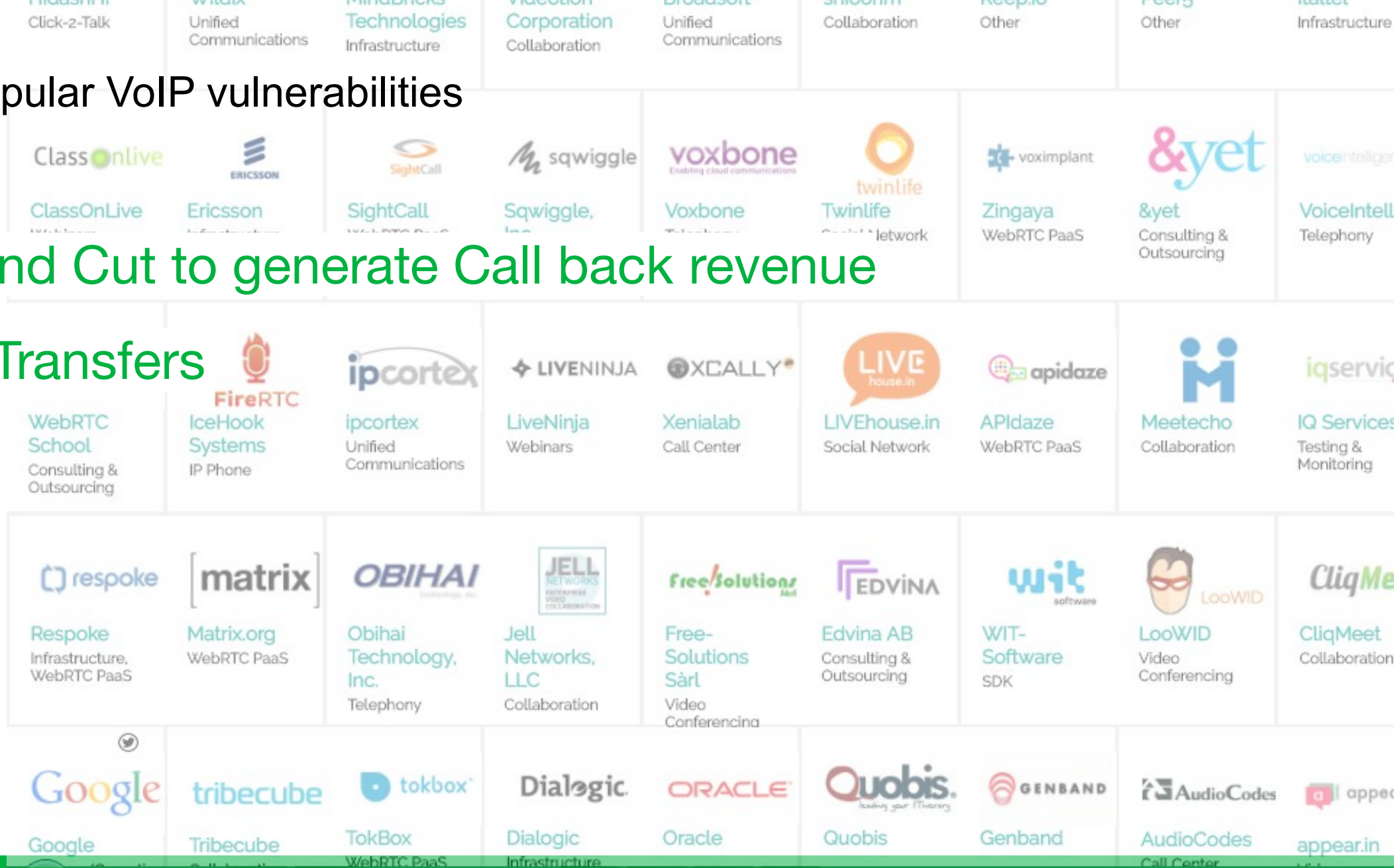
Blind Call Transfers

Call Cards

Vishing

VOMIT

SPIT



Detection of Fraud and Countermeasures

Call signatures

Risk Assessment

Fraud occur in off-hours

- when networks are often monitored less closely so that they can go unnoticed longer

Back propagation Neural Network to detect SPAM calls

VoIP Intrusion Detection

Aggregate data from honeypot application and traffic monitoring to ANN



Recognizing attacks using ANN

Classifying Possible Intrusions

options tests; options scanning; call testing; unknown protocol; register and call; registration test, registration flooding; register attempt

Aggregate data from honeypot application and traffic monitoring to design response



References

- <https://telecom.altanai.com/>
 - [VOIP Call Metric Monitoring and MOS \(Mean Opinion Score\)](#)
 - [OTT \(Over the Top \) Communication applications](#)
 - [CLI/NCLI, Robocalls and STIR/SHAKEN](#)
- [Voice Activity Detection - Link Kaggle -https://www.kaggle.com/voice-activity-detection-with-webrtc](#)
- [Telecom Churn Prediction - Link kaggle -https://www.kaggle.com/altanai/telecom-churn-prediction](#)
- QoS Evaluation Based on Extend E-Model in VoIP Hongli Zhangab, Zhimin Gua , Zhenqing Tianb a School of Computer Science and Technology, Beijing Institute of Technology b Media College, Inner Mongolia Normal University ,China
- Voice quality prediction models and their application in VoIP networks - September 2006IEEE Transactions on Multimedia DOI: 10.1109/TMM.2006.876279 , Lingfen Sun, Emmanuel Ifeachor
- TransNexus TELECOM FRAUD CALL SCENARIOS
- Nassar, Mohamed & State, Radu & Festor, Olivier. (2007). VoIP Honeypot Architecture. 109 - 118. 10.1109/INM.2007.374775.



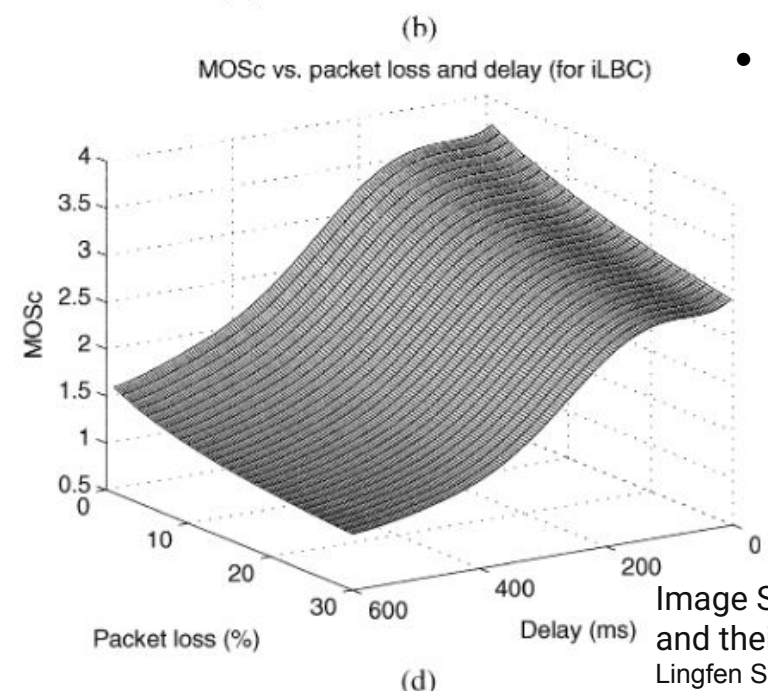
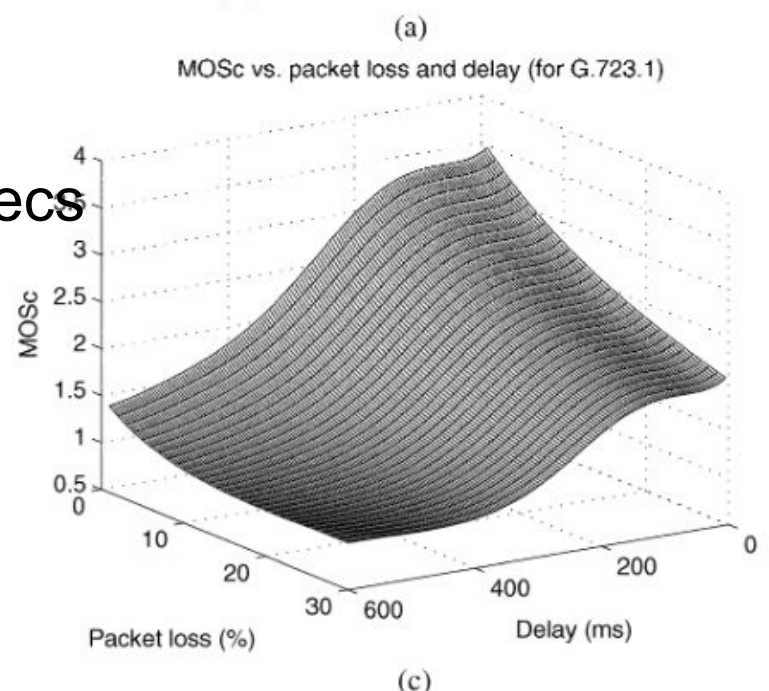
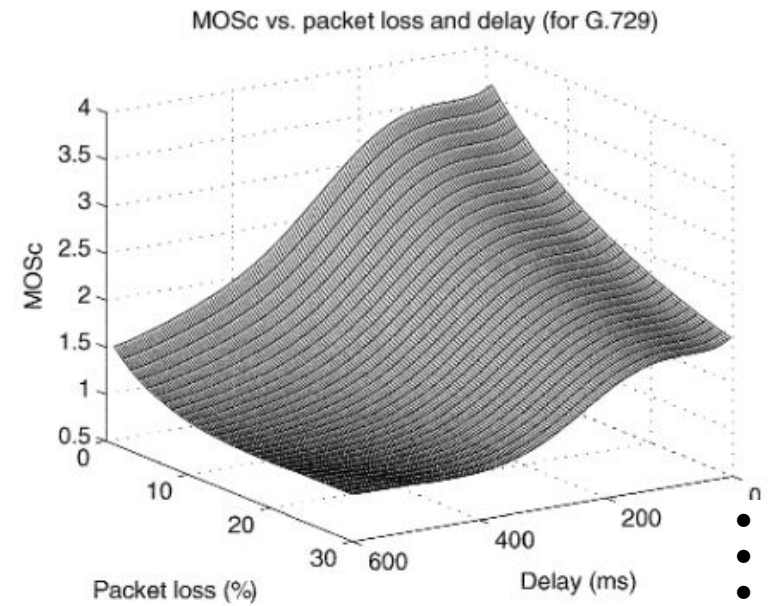
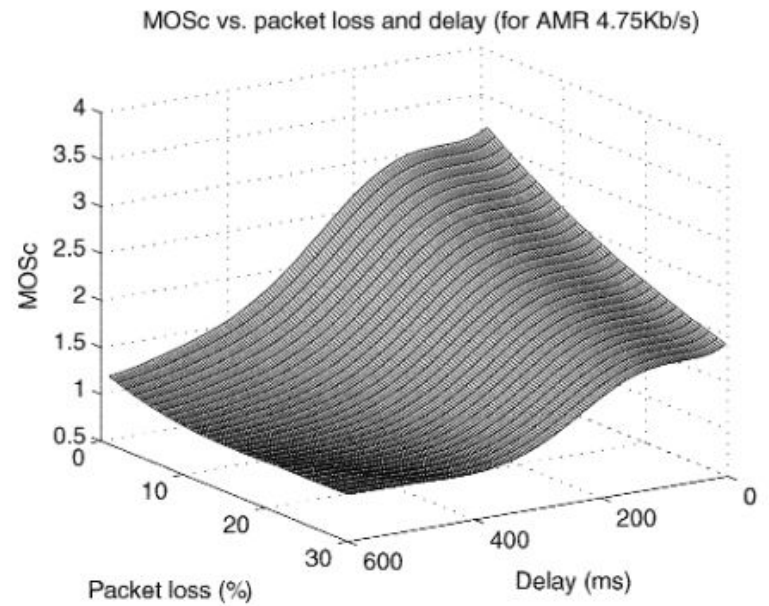
Thank you :)



Extras



Performance Metrics of Packet Loss on Different Codecs



- G.729 (8 Kb/s),
- G.723.1 (6.3 Kb/s),
- AMR(the highest mode, 12.2 Kb/s and the lowest, 4.75 Kb/s)
- iLBC (15.2 Kb/s).

Fig. 12. MOSc versus packet loss and delay for different codecs: (a) for AMR (4.75 Kb/s); (b) for G.729; (c) for G.723.1; (d) for iLBC.

Image Source : Voice quality prediction models and their application in VoIP networks
Lingfen Sun , Emmanuel Ifeakor

