

LTE Virtualization: from Theoretical Gain to Practical Solution

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Presented by:

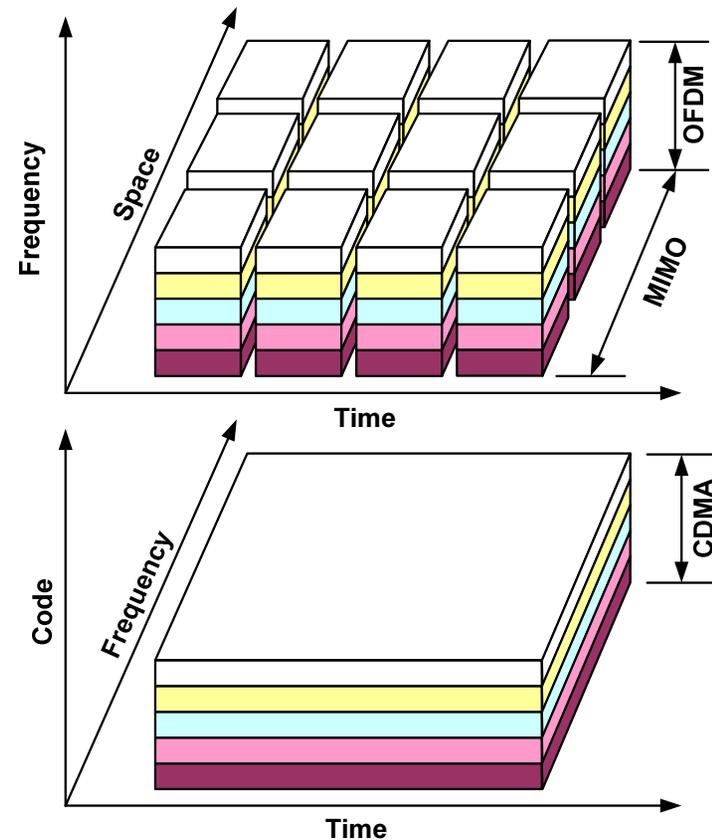
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- ▶ Radio resources are scarce
- ▶ Instant traffic load of different operators can be different
- ▶ Resource sharing (spectrum sharing) can enhance the resource utilization and improve the system performance
- ▶ Spectrum sharing in the context of network virtualization
 - Wireless virtualization
 - LTE virtualization
- ▶ LTE virtualization makes sense
 - Reducing the CAPEX of operators
 - Improving system performance
 - Saving energy
 - OFDM-based PHY technique
 - ...

- ▶ Introduction to Network Virtualization
 - General Concept of Network Virtualization
 - Wireless Virtualization
 - Our Vision on LTE virtualization
- ▶ Analytical Investigation on Virtualization Gain
 - Analytical Approach based on Queue Theory
 - Validation by Simulation
- ▶ Multi-party Spectrum Sharing Model for LTE Virtualization
- ▶ Load Estimation based on Realtime Services
 - Idea → Improvement → Results
- ▶ Conclusion

- ▶ Network Virtualization: multiple virtual networks running on a common physical network
 - Abstraction, isolation, sharing
 - Node virtualization and link virtualization
 - E.g. 4WARD project [1]
- ▶ Wireless Virtualization
 - A nature extension from wired to wireless virtualization
 - Virtualization of the wireless resources on the air interface is one scheduling problem of Tx/Rx power, frequency, time, and code or space allocation, like well known wireless transmission strategies: FDMA, TDMA, CDMA and SDMA
 - Issue: the wireless links suffer more interference than the wired ones



- ▶ The Virtual Operator (VO) of LTE Virtualization is different from the traditional MNVO
 - VO has the control of his network from end-to-end
 - VO has more flexibility of upgrading, expanding and tuning his network
 - VO pays less money to run a mobile network
 -
- ▶ Our solution focuses on the air interface of LTE
 - Spectrum sharing in the context of LTE virtualization
 - Revealing the potential advantage of LTE virtualization
 - Proposing solutions on some of the aspects for applying it
 - Hardware virtualization is out of our scope. See more [2]

- ▶ Starting from a FTP-only scenario to reveal the gain of LTE virtualization
 - FTP is best effort
 - One FTP connection can theoretically occupy the whole capacity of one cell
- ▶ M/G/1-PS can be employed to model this scenario
 - Each flow in the system has the ability to fully utilize the whole capacity
 - **M** requires the flow arrival follows a Poisson process with arrival rate λ
 - **G** means a generally distributed service time
 - **1** server applies the Processor Sharing (**PS**) discipline
- ▶ Then,
 - The expected sojourn time t (average FTP download time of a file size x) [3]

$$E_{M/G/1-PS}\{t\} = \frac{x}{C} \cdot \left(\frac{1}{1-\rho} \right) \quad \text{where, } \rho = \frac{\lambda \cdot \bar{x}}{C}$$

- If “n” virtual operators are sharing the spectrum in on cell, then the capacity C and arrival rate λ increase ”n” times

$$\rho = \frac{n \cdot \lambda \cdot \bar{x}}{n \cdot C} = \frac{\lambda \cdot \bar{x}}{C}; \quad \rightarrow \quad E_{M/G/1-PS}\{t\} = \frac{x}{n \cdot C} \cdot \left(\frac{1}{1-\rho} \right)$$

▶ Temporary conclusion and questions

- Large multiplexing gain can be obtained by the spectrum sharing between VOs
- Question 1: does this hold for simulations?
- Question 2: for LTE systems, each VoIP connection requires at least one PRB to transmit packets. If a lot of VoIP users are available, is this gain still obvious?

TABLE I
FTP DOWNLOADING

Parameter	Assumption
average file size $\bar{\chi}$	exponential distribution with mean of 5 [MB]
average IAT $1/\lambda$	exponential distribution

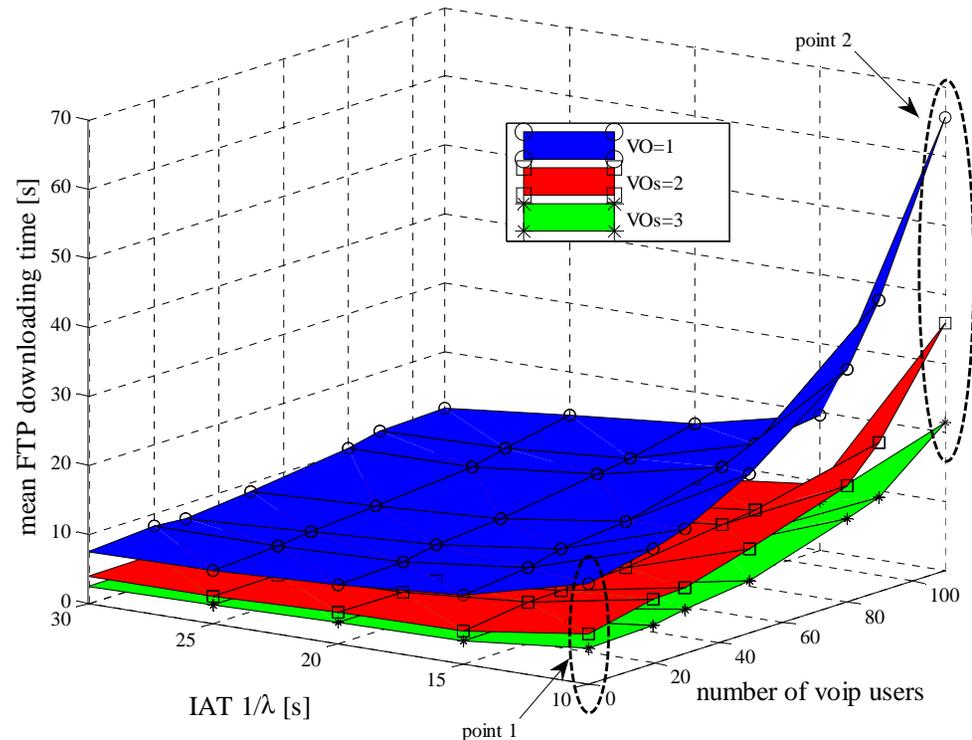
TABLE II
VOIP TRAFFIC MODEL

Parameter	Assumption
activity factor	0.5
talking time	exponential distribution with mean of 3 [s]
silence time	exponential distribution with mean of 3 [s]
call duration	complete run time
data packet size	40 [byte] every 20 [ms] during talking time
SID packet	20 [byte] every 160 [ms] during silence

TABLE III
SIMULATION MODELS

Parameter	Assumption
number of VO	1 to 3
eNBs in each VO	19 hexagonally in 3 tiers
sectors of each eNB	3
site-2-site distance	500 [m]
spectrum of each VO	5 MHz := 25 PRBs
total output power	20 [W] := 43 [dBm]
path loss model	COST 231
2D slow fading	20 m decorrelation distance
fast fading	Rosa Zheng model [12]
mobility model	reallocation of stationary users
traffic model	Table I and II
scheduling	VoIP-prioritized Round-Robin

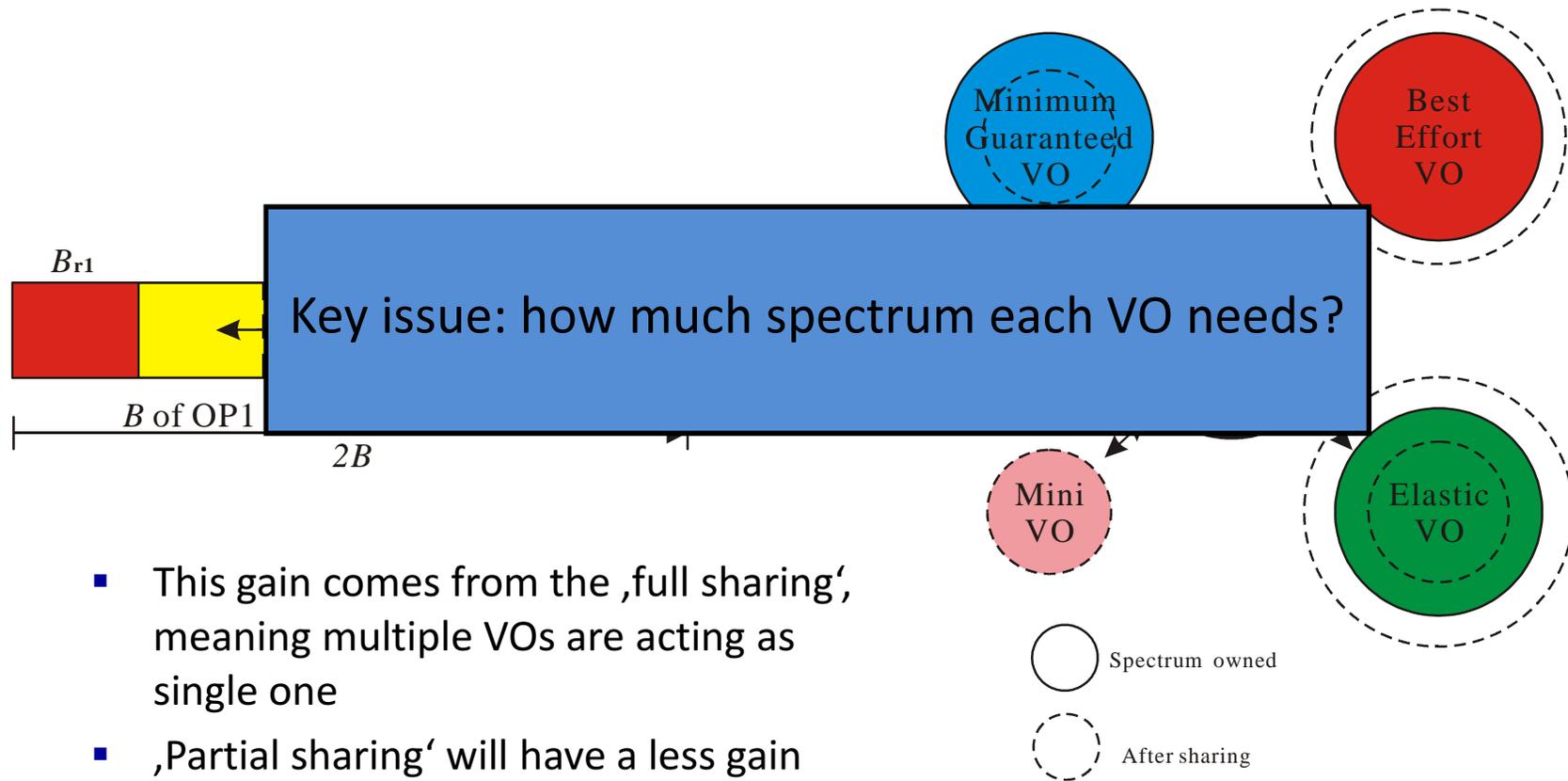
- ▶ The benchmark is the average FTP download time
- ▶ The results from simulations are closely matched to analytical analysis
- ▶ Even many VoIP users are in the system, the gain from spectrum sharing is still obvious



Points	VOs	M/G/1-PS	Simulation
point 1	$O = 1 \rightarrow 2$	50%	$1-5.91s/11.91s = 51.4\%$
FTP only	$O = 1 \rightarrow 3$	66.7%	$1-3.92s/11.91s = 67.1\%$
point 2	$O = 1 \rightarrow 2$	50%	$1-35.9s/66.71s = 46.2\%$
FTP and VoIP=100	$O = 1 \rightarrow 3$	66.7%	$1-21.5s/65.71s = 67.8\%$

▶ Sharing model:

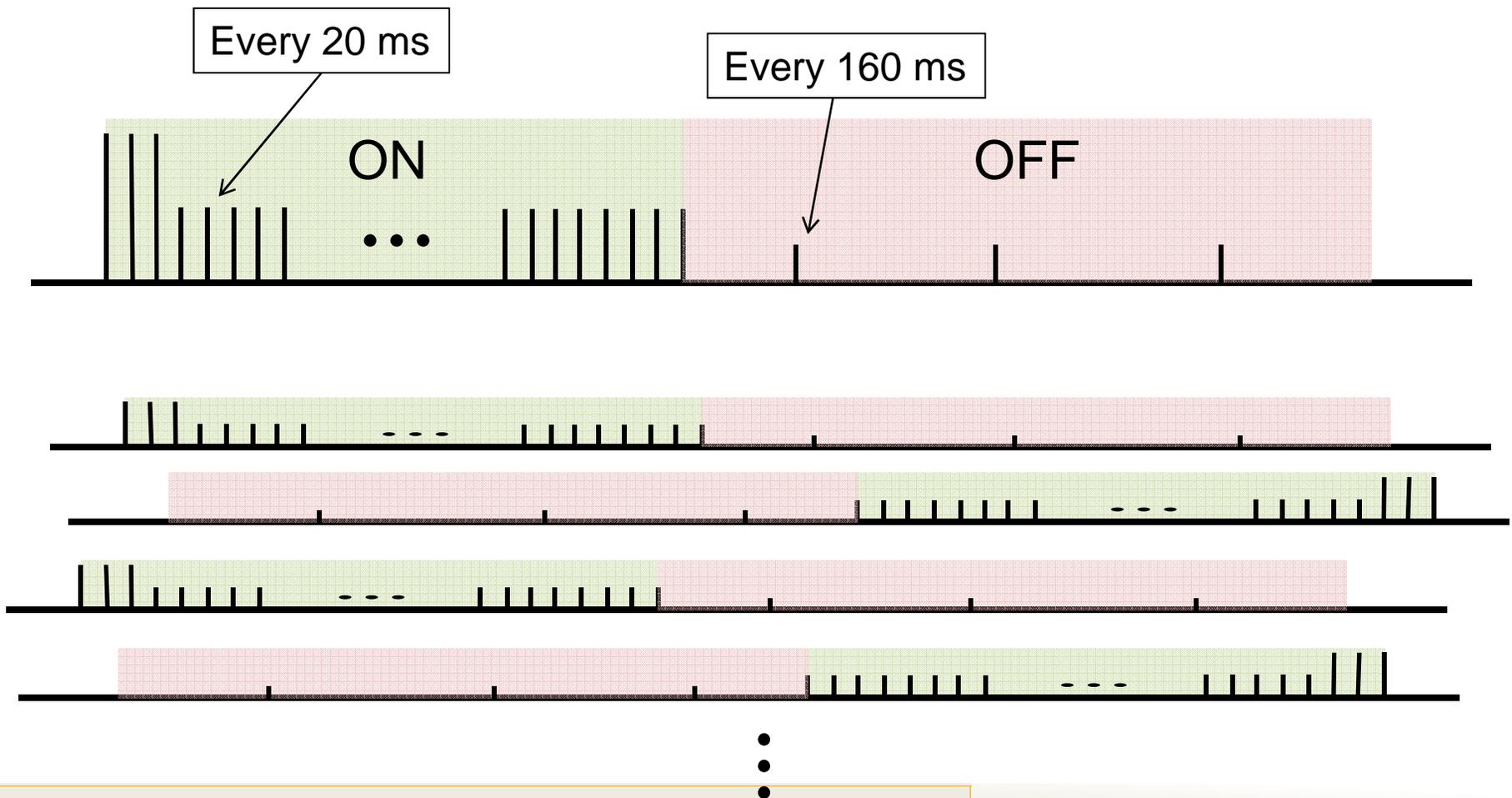
- Illustration of the spectrum sharing among multiple VOs
- Different VOs may have different strategies

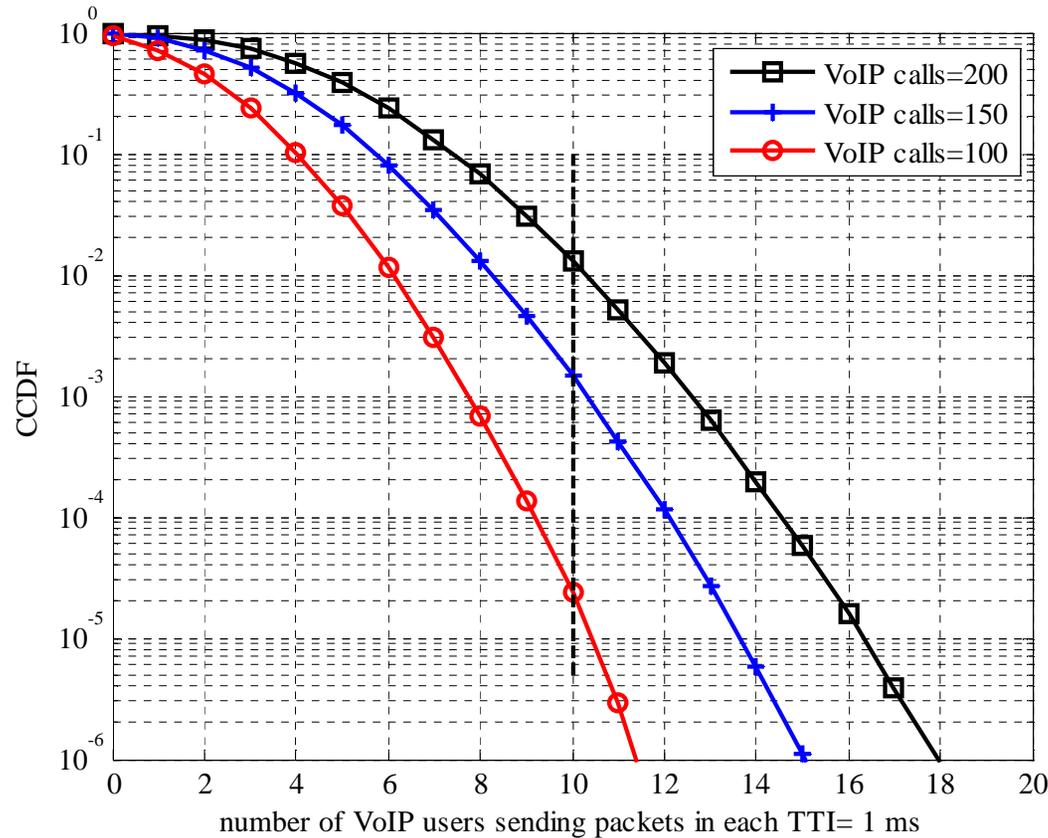


- This gain comes from the 'full sharing', meaning multiple VOs are acting as single one
- 'Partial sharing' will have a less gain

- ▶ Real time services; GBR; CBR ... normally require constant bit rates
- ▶ Real time services should have higher priority than non-real time services to guarantee QoS
- ▶ VoIP is one common but one of the most services of operators
- ▶ idea
 - Number VoIP users →
 - Number of active VoIP users →
 - Number of VoIP packets →
 - Required BW (PRBs) →
 - Load estimation (including some margin) →
 - Spectrum Sharing

- Length of ON and OFF periods follows an exponential distribution with a same mean

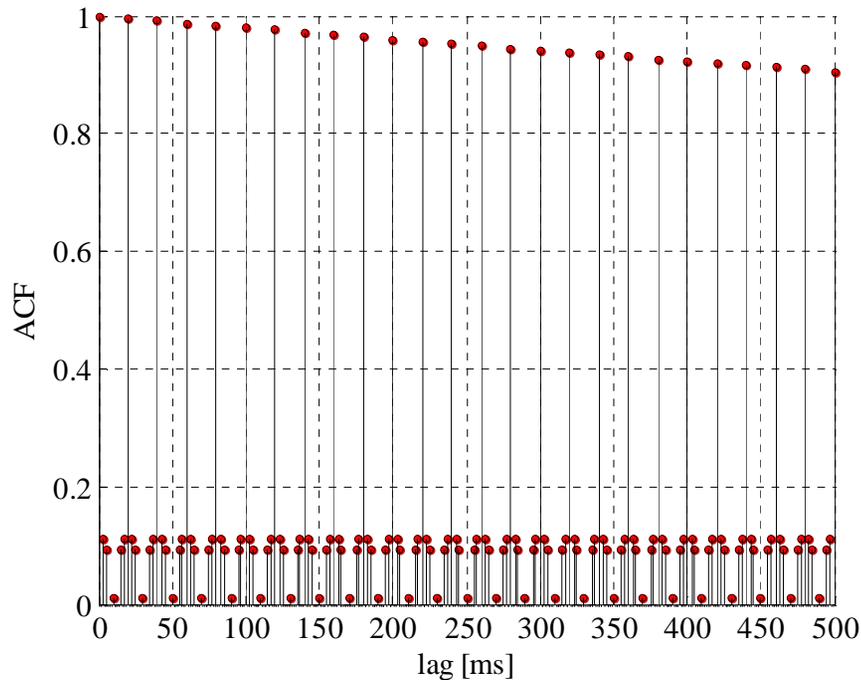




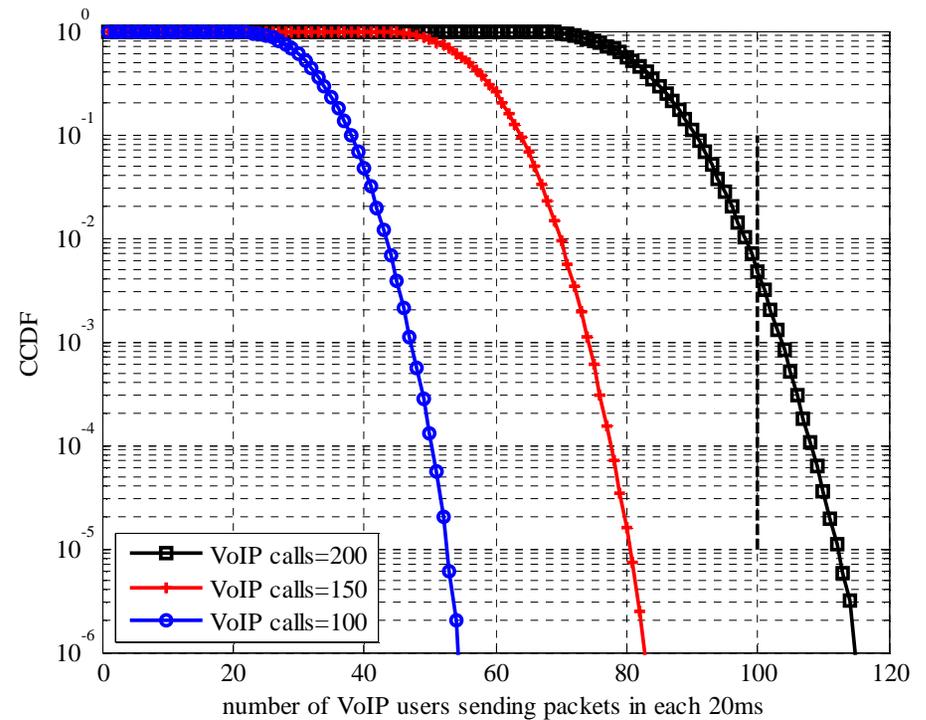
Number of active VoIP users in each TTI = 1ms

→ 10 user/ms can cover > 99% ?

- ▶ Improving based on 20ms estimation
 - From the traffic model of VoIP, 1 packet every 20 ms in ON period
 - Checking the traffic characteristics in every 20ms
 - Load estimation can be more accurate

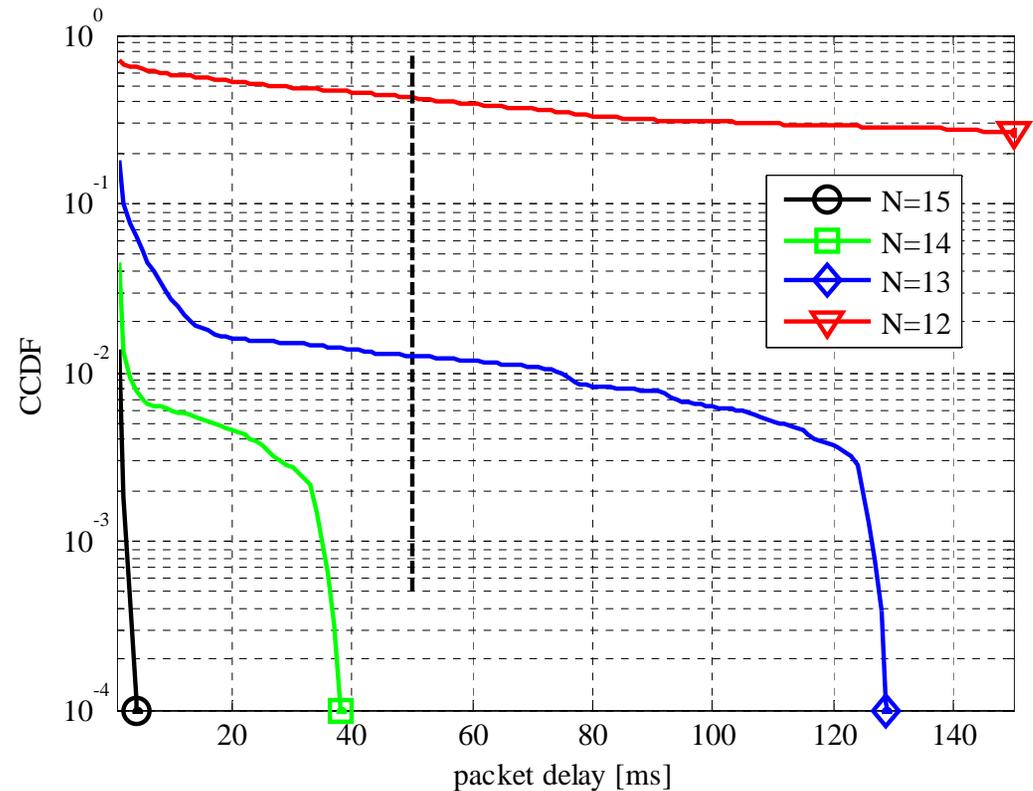


Autocorrelation of user traffic for 200 VoIP calls
Shows a strong correlation every 20 ms



Traffic load in every 20ms:
100 users/20ms = 5 users/ms is enough!

- ▶ Each active user has one packet of 40 Bytes
- ▶ In our scenario, each 40 byte packet averagely requires 2-3 PRBs to transmit
- ▶ Packet delay is selected as the benchmark, and 50 ms is the delay budget
- ▶ 13-14 PRBs are enough to support the data transmission of 200 VoIP calls, **but this number is scenario-specific**
- ▶ Spared PRBs can be shared between different VOs



Validation of load estimation
Delay performance of 200 VoIP calls
N: number of PRBs reserved

- ▶ Large multiplexing gain can be obtained via the spectrum sharing of LTE virtualization
 - Analytical observation
 - Simulative validation
- ▶ Reasonable spectrum sharing model and strategies are necessary
- ▶ Load estimation is the key issue for the success of the spectrum sharing
 - The load estimation based on the real time traffic model is accurate, easy and insensitive to the parameters
- ▶ Further works
 - Partially overlapping cell scenario
 - Heterogeneous traffic models

THANK YOU!

ANY QUESTION??

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