Современное отечественное решение ВКС Vinteo

VINTEO

Video Communication Core

Интеграция Vinteo с инфраструктурой зарубежных производителей: Cisco, Polycom, Microsoft

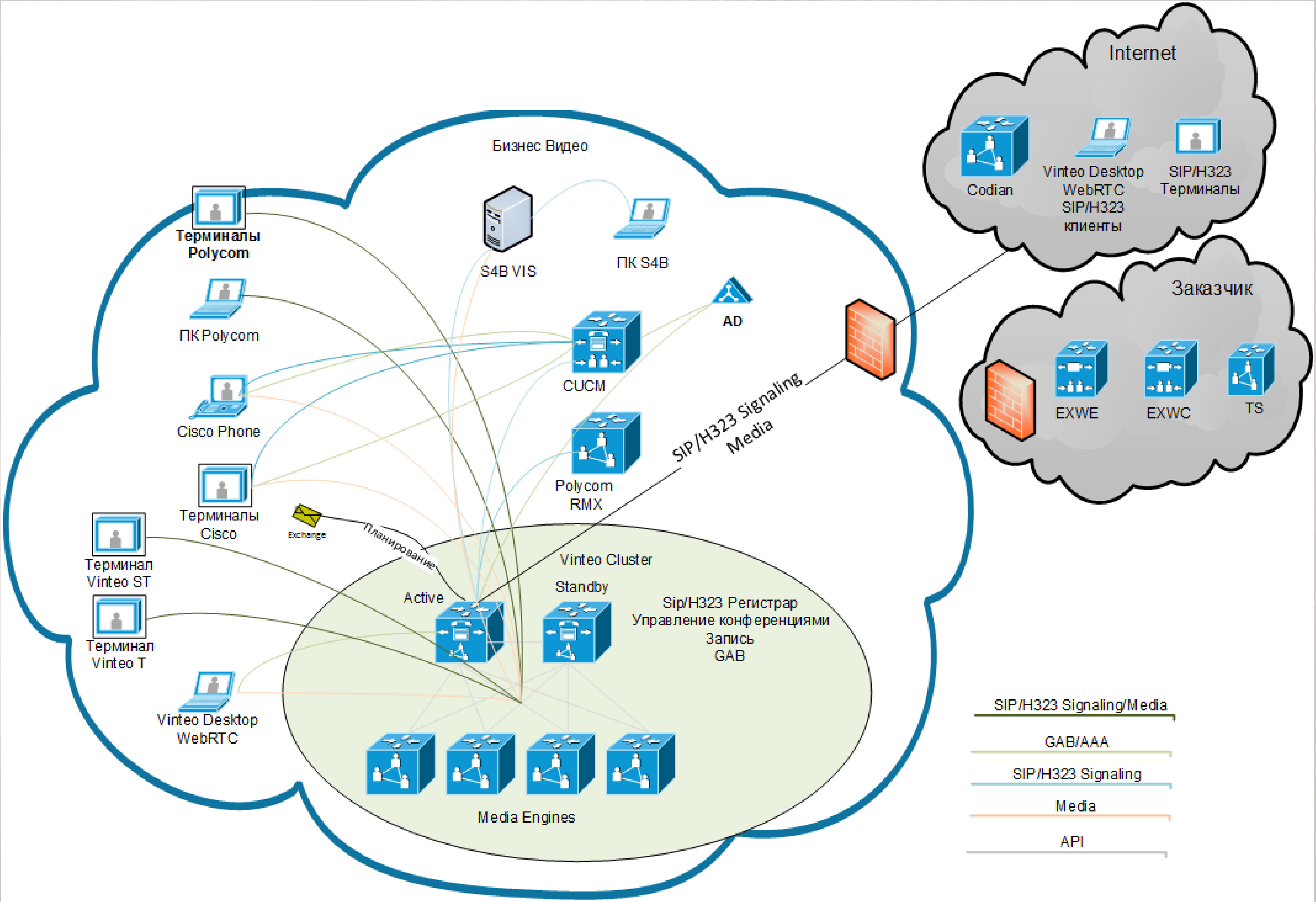
Видеорешения без компромиссов

Теперь ВКС подстраивается под вас

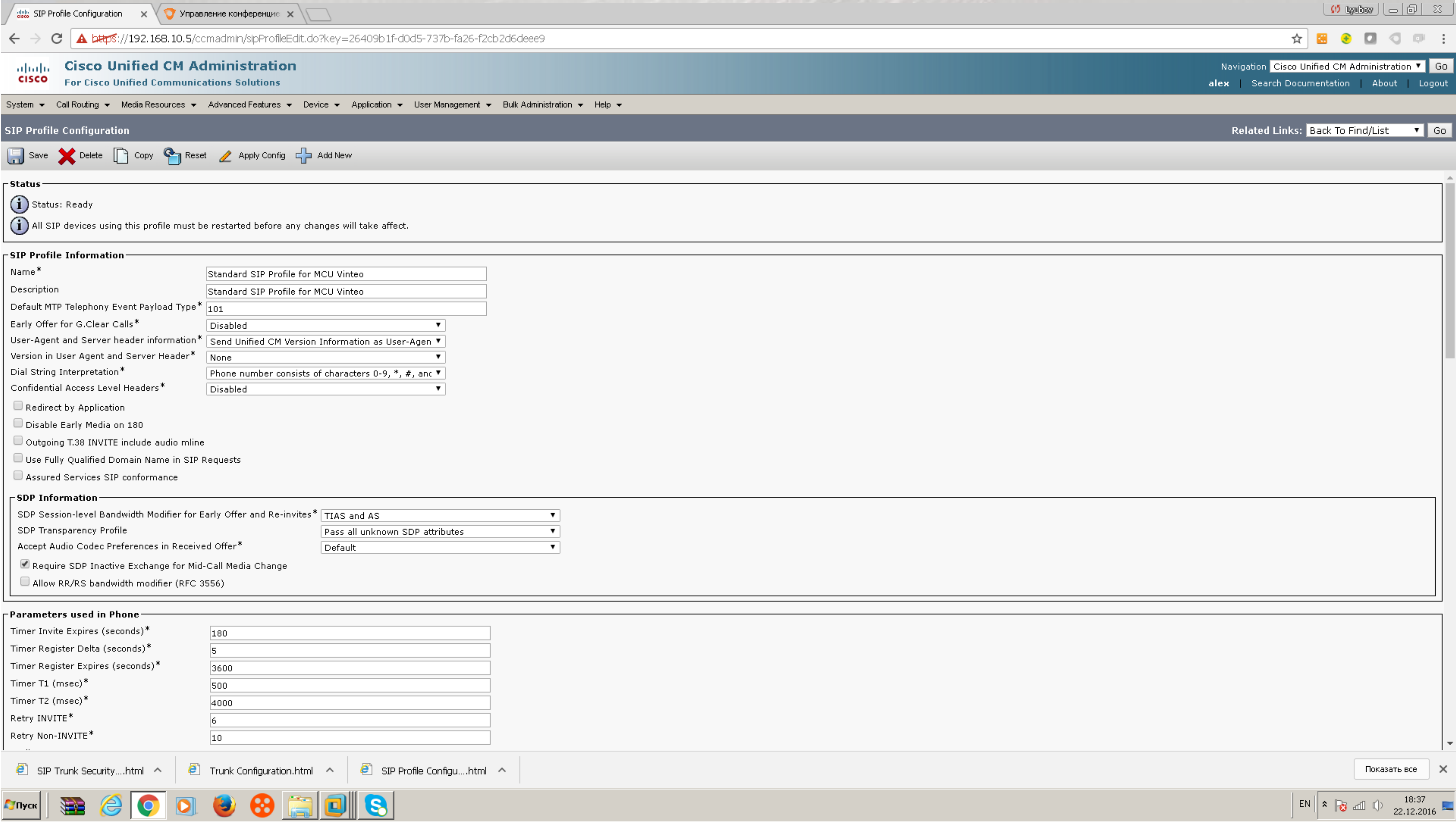
Интеграция

- Общая схема интеграции Vinteo с системами зарубежных производителей
- Cisco Codian & TMS
- Cisco ExpE & ExwC
- Cisco CUCM & CMS
- Polycom (Plantronics)
- S4B
- Сторонние продукты
- LDAP
- Запись и трансляция

- Общая схема интеграции Vinteo с системами зарубежных производителей



• Cisco CUCM SIP Profile



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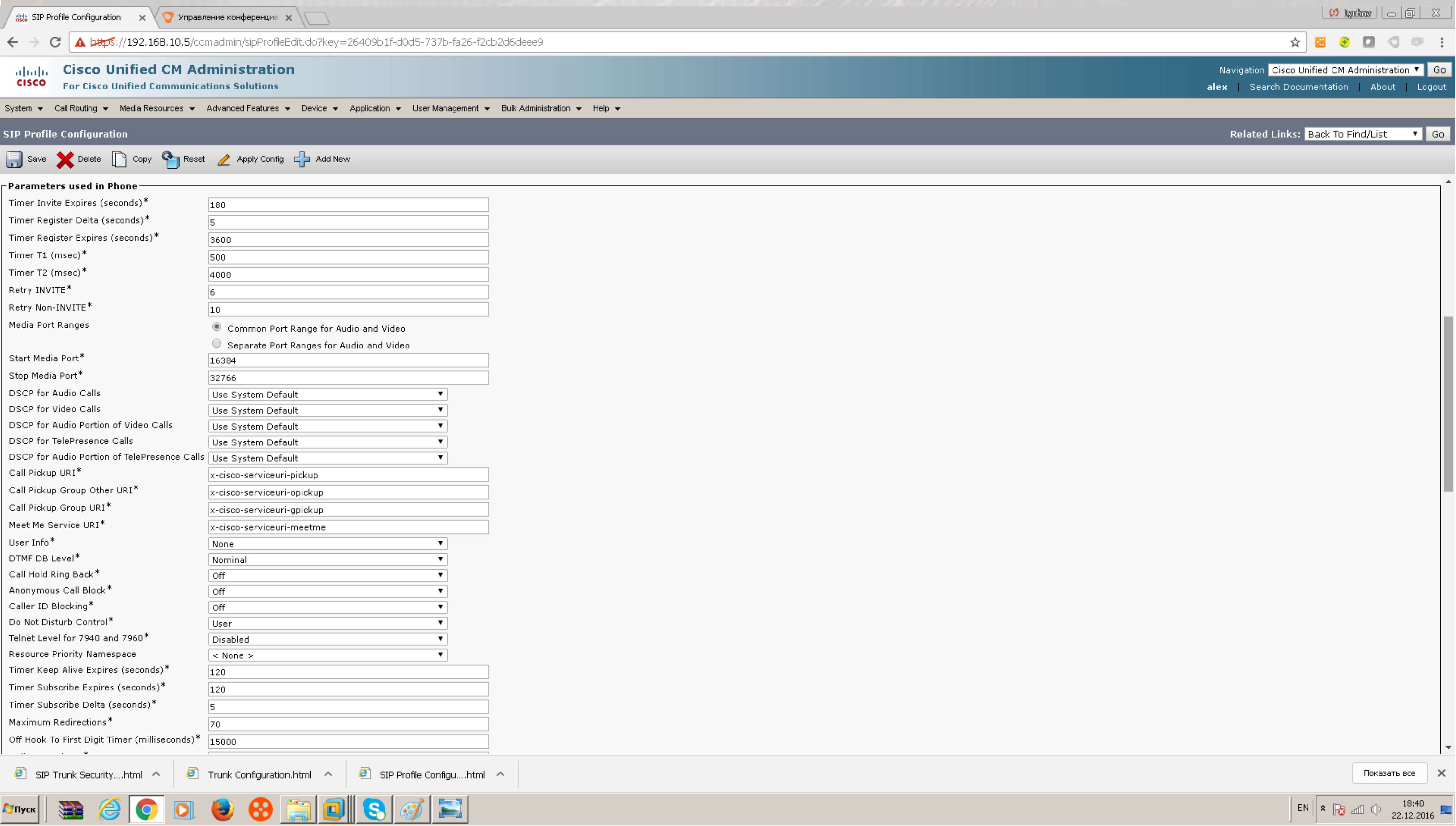

SIP Profile Configuration
Управление конференцие
https://192.168.10.5/ccmadmin/sipProfileEdit.do?key=26409b1f-d0d5-737b-fa26-f2cb2d6deee9
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SIP Profile Configuration
Related Links: Back To Find/List Go
Save Delete Copy Reset Apply Config Add New
Parameters used in Phone
Timer Invite Expires (seconds)* 180
Timer Register Delta (seconds)* 5
Timer Register Expires (seconds)* 3600
Timer T1 (msec)* 500
Timer T2 (msec)* 4000
Retry INVITE* 6
Retry Non-INVITE* 10
Media Port Ranges
Common Port Range for Audio and Video
Separate Port Ranges for Audio and Video
Start Media Port* 16384
Stop Media Port* 32766
DSCP for Audio Calls Use System Default
DSCP for Video Calls Use System Default
DSCP for Audio Portion of Video Calls Use System Default
DSCP for TelePresence Calls Use System Default
DSCP for Audio Portion of TelePresence Calls Use System Default
Call Pickup URI* x-cisco-serviceuri-pickup
Call Pickup Group Other URI* x-cisco-serviceuri-opickup
Call Pickup Group URI* x-cisco-serviceuri-gpickup
Meet Me Service URI* x-cisco-serviceuri-meetme
User Info* None
DTMF DB Level* Nominal
Call Hold Ring Back* Off
Anonymous Call Block* Off
Caller ID Blocking* Off
Do Not Disturb Control* User
Telnet Level for 7940 and 7960* Disabled
Resource Priority Namespace < None >
Timer Keep Alive Expires (seconds)* 120
Timer Subscribe Expires (seconds)* 120
Timer Subscribe Delta (seconds)* 5
Maximum Redirections* 70
Off Hook To First Digit Timer (milliseconds)* 15000
SIP Trunk Security....html
Trunk Configuration.html
SIP Profile Configu....html
Показать все
Пуск
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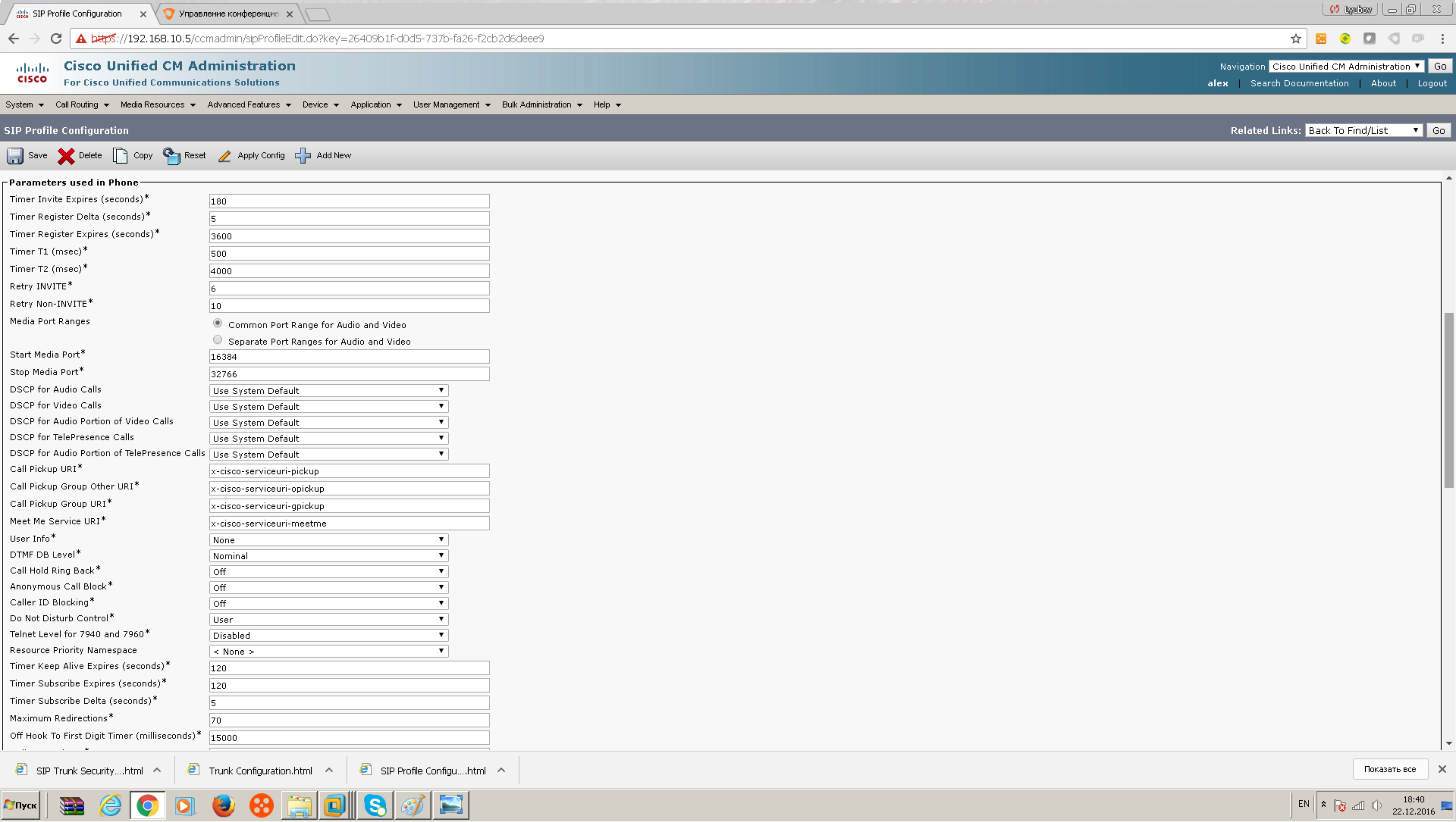
18:40
22.12.2016

- # Cisco CUCM



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Call Pickup Group URI* x-cisco-serviceuri-gpickup
Meet Me Service URI* x-cisco-serviceuri-meetme
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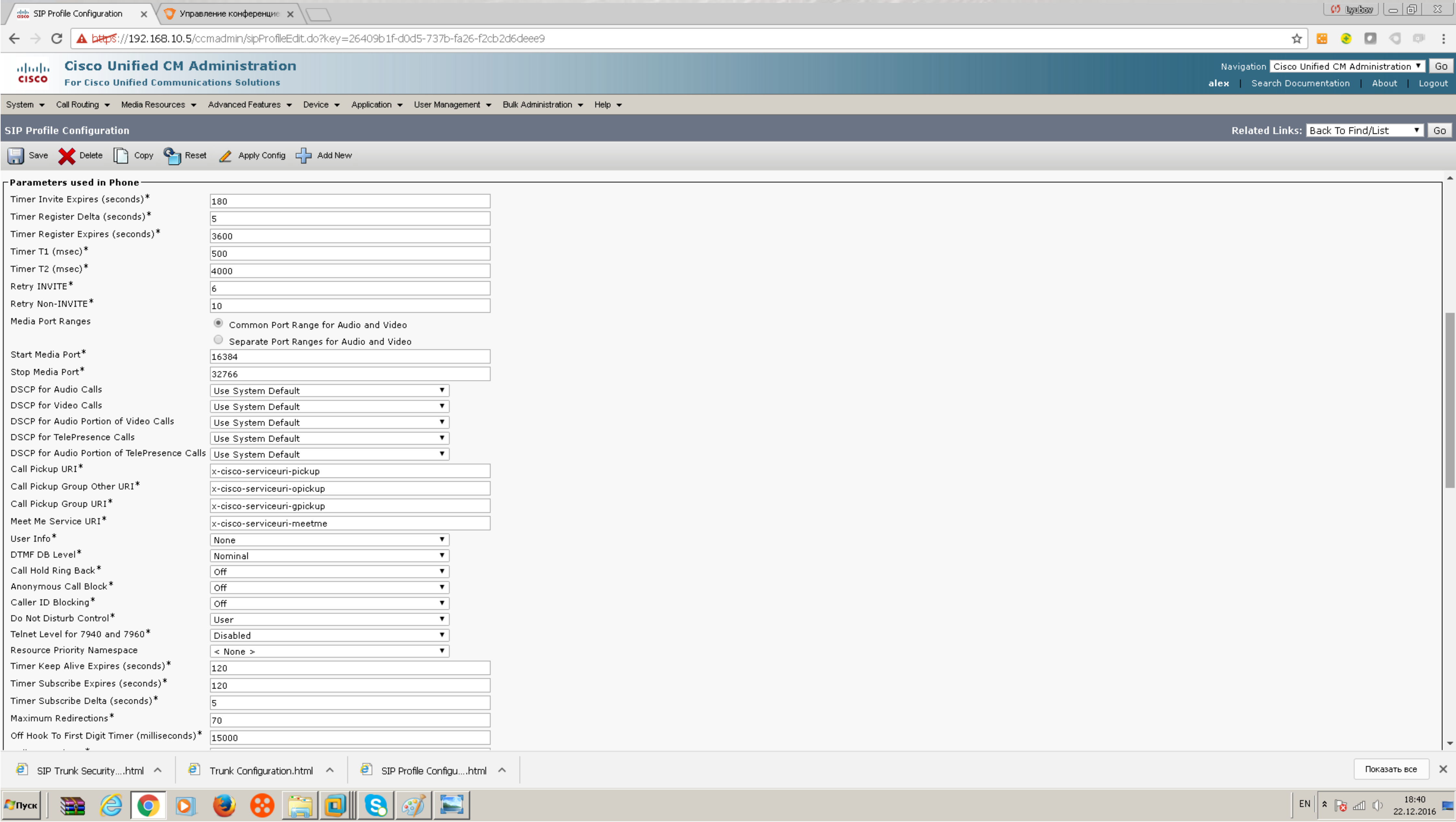
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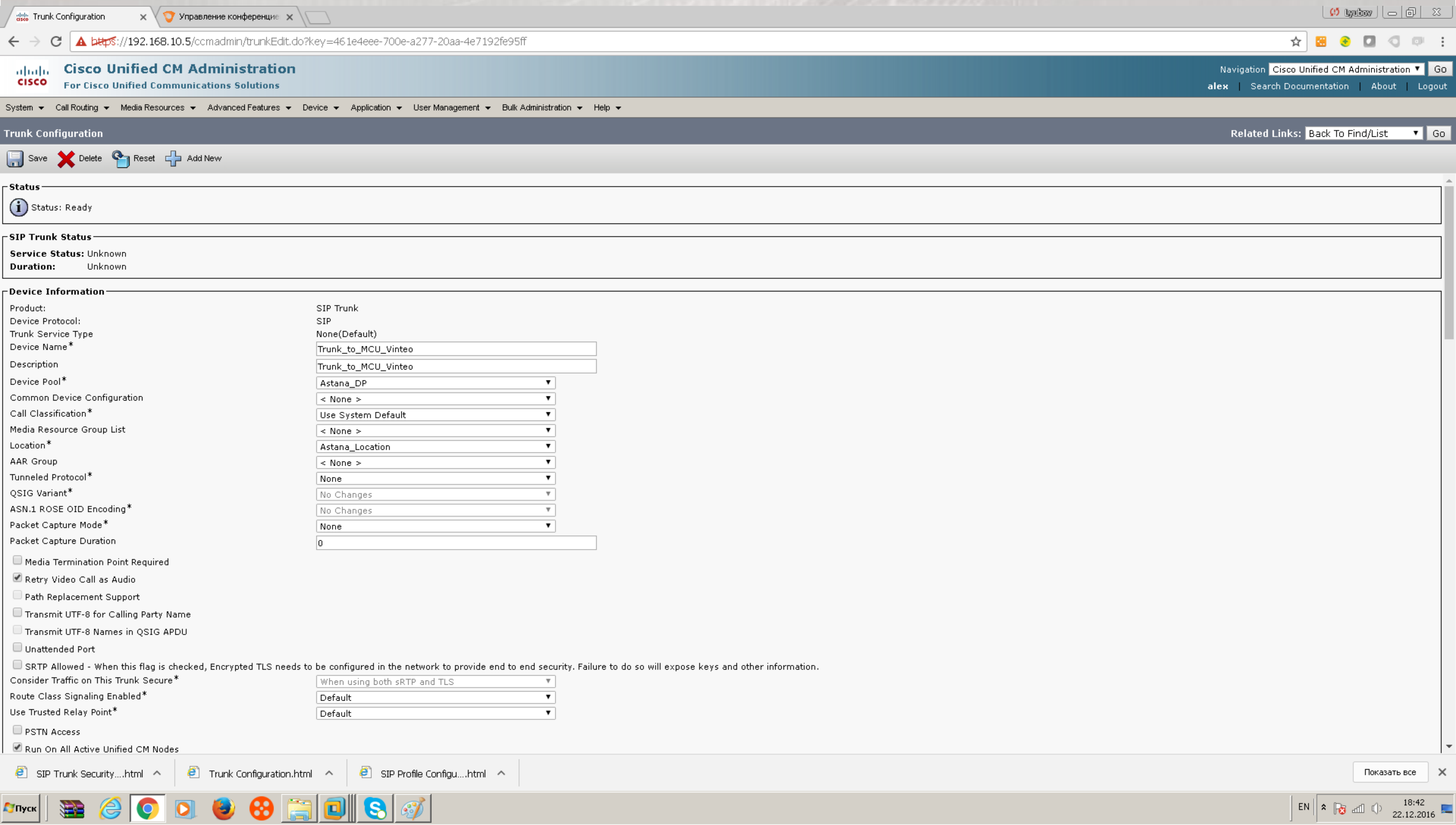
• Cisco CUCM SIP Trunk



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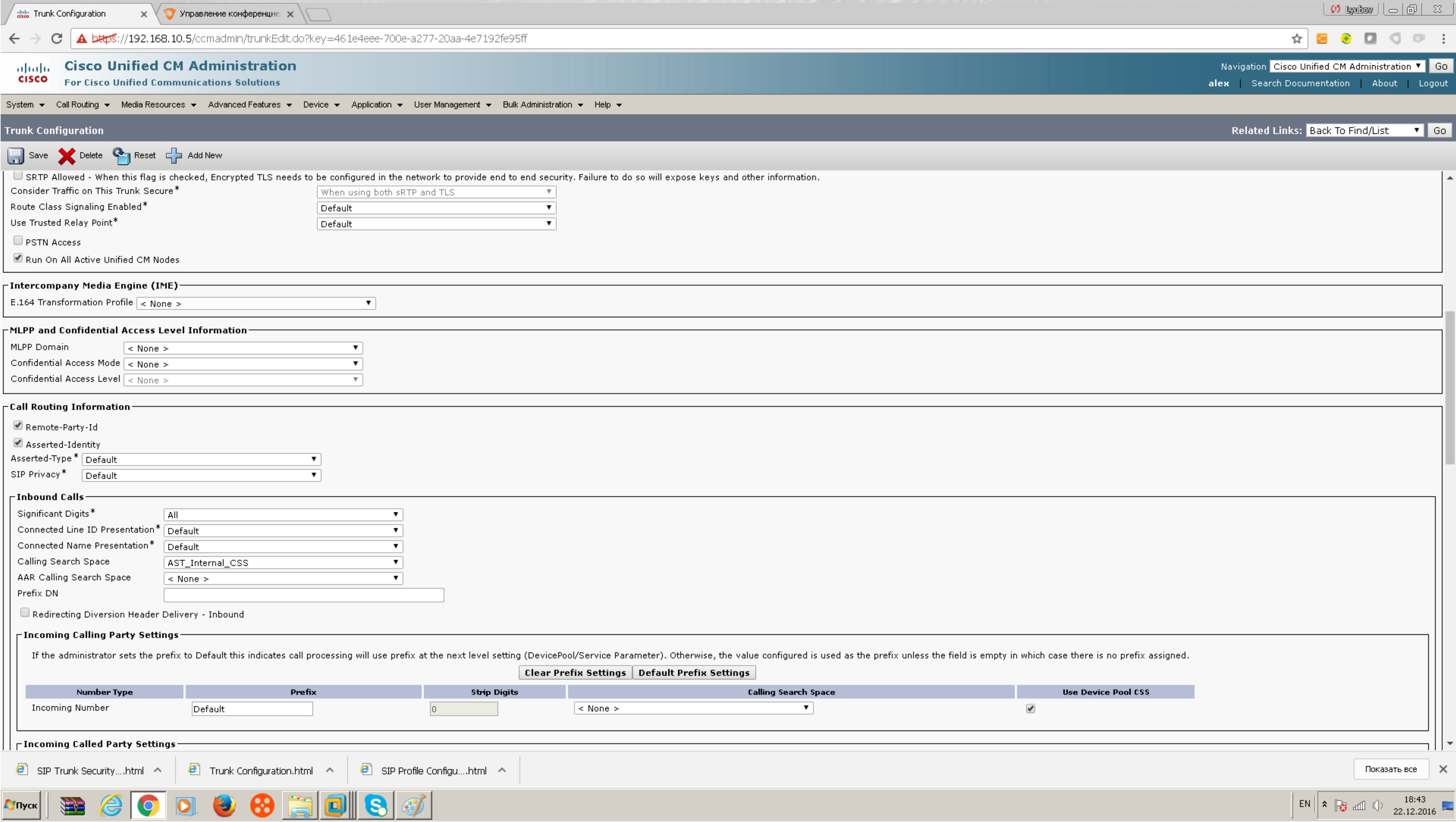

Trunk Configuration
Управление конференциями
https://192.168.10.5/ccmadmin/trunkEdit.do?key=461e4eee-700e-a277-20aa-4e7192fe95ff
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Trunk Configuration
Related Links: Back To Find/List Go
Save Delete Reset Add New
SRTP Allowed - When this flag is checked, Encrypted TLS needs to be configured in the network to provide end to end security. Failure to do so will expose keys and other information.
Consider Traffic on This Trunk Secure* When using both sRTP and TLS
Route Class Signaling Enabled* Default
Use Trusted Relay Point* Default
PSTN Access
Run On All Active Unified CM Nodes
Intercompany Media Engine (IME)
E.164 Transformation Profile < None >
MLPP and Confidential Access Level Information
MLPP Domain < None >
Confidential Access Mode < None >
Confidential Access Level < None >
Call Routing Information
Remote-Party-Id
Asserted-Identity
Asserted-Type* Default
SIP Privacy* Default
Inbound Calls
Significant Digits* All
Connected Line ID Presentation* Default
Connected Name Presentation* Default
Calling Search Space AST_Internal_CSS
AAR Calling Search Space < None >
Prefix DN
Redirecting Diversion Header Delivery - Inbound
Incoming Calling Party Settings
If the administrator sets the prefix to Default this indicates call processing will use prefix at the next level setting (DevicePool/Service Parameter). Otherwise, the value configured is used as the prefix unless the field is empty in which case there is no prefix assigned.
Clear Prefix Settings Default Prefix Settings
Number Type Prefix Strip Digits Calling Search Space Use Device Pool CSS
Incoming Number Default 0 < None >
Incoming Called Party Settings
SIP Trunk Security....html
Trunk Configuration.html
SIP Profile Configu....html
Показать все
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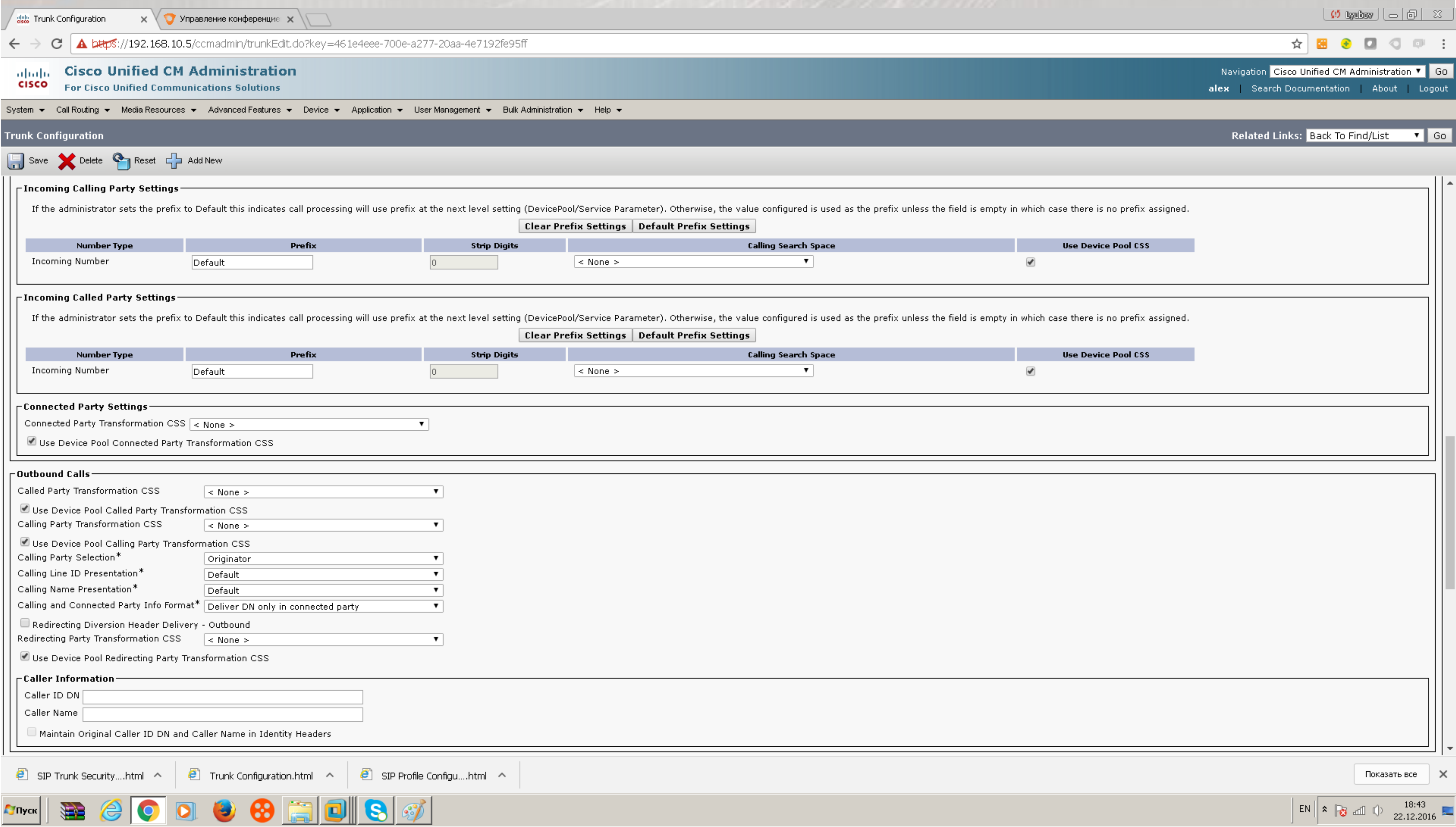

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Clear Prefix Settings Default Prefix Settings
Number Type Prefix Strip Digits Calling Search Space Use Device Pool CSS
Incoming Number Default 0 < None >
Connected Party Settings
Connected Party Transformation CSS < None >
Use Device Pool Connected Party Transformation CSS
Outbound Calls
Called Party Transformation CSS < None >
Use Device Pool Called Party Transformation CSS
Calling Party Transformation CSS < None >
Use Device Pool Calling Party Transformation CSS
Calling Party Selection* Originator
Calling Line ID Presentation* Default
Calling Name Presentation* Default
Calling and Connected Party Info Format* Deliver DN only in connected party
Redirecting Diversion Header Delivery - Outbound
Redirecting Party Transformation CSS < None >
Use Device Pool Redirecting Party Transformation CSS
Caller Information
Caller ID DN
Caller Name
Maintain Original Caller ID DN and Caller Name in Identity Headers
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Trunk Configuration.html
SIP Profile Configu....html
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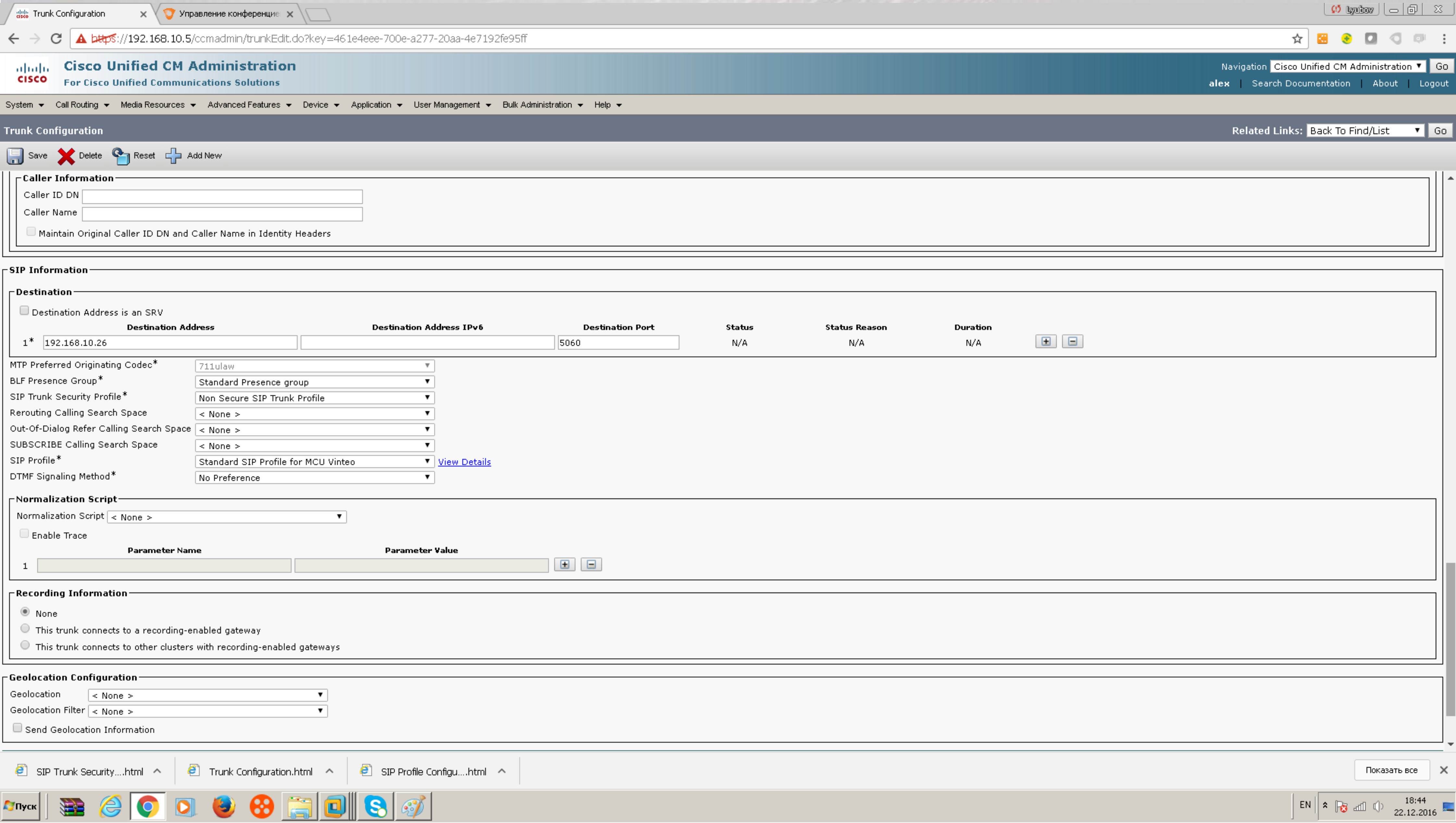
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Caller Information
Caller ID DN
Caller Name
Maintain Original Caller ID DN and Caller Name in Identity Headers
SIP Information
Destination
Destination Address is an SRV
Destination Address
Destination Address IPv6
Destination Port
Status
Status Reason
Duration
1* 192.168.10.26 5060 N/A N/A N/A
MTP Preferred Originating Codec* 711ulaw
BLF Presence Group* Standard Presence group
SIP Trunk Security Profile* Non Secure SIP Trunk Profile
Rerouting Calling Search Space < None >
Out-Of-Dialog Refer Calling Search Space < None >
SUBSCRIBE Calling Search Space < None >
SIP Profile* Standard SIP Profile for MCU Vinteo View Details
DTMF Signaling Method* No Preference
Normalization Script
Normalization Script < None >
Enable Trace
Parameter Name
Parameter Value
1
Recording Information
None
This trunk connects to a recording-enabled gateway
This trunk connects to other clusters with recording-enabled gateways
Geolocation Configuration
Geolocation < None >
Geolocation Filter < None >
Send Geolocation Information
SIP Trunk Security....html
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SIP Profile Configu....html
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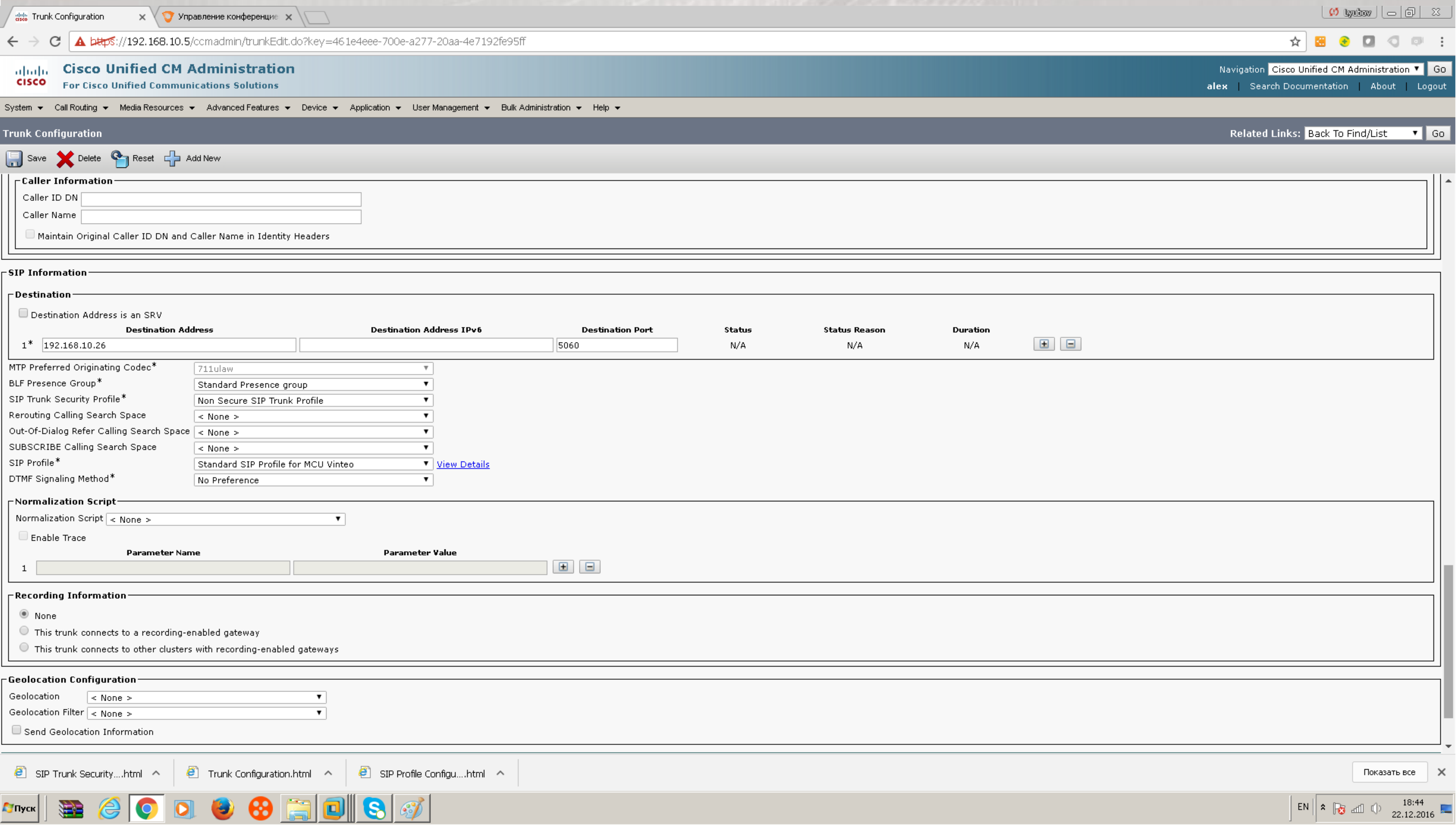
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Вопросы?

Барышев Михаил
mkl@vinteo.ru
+7985-920-6287
8-800-333-40-16
vinteo.com