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June 18-19



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At WebRTC Conference & Expo IV, you will get the most up to date, technically sound advice and view the latest platforms & applications that will show you how to harness the power of WebRTC. Implement real time solutions across a multitude of applications - choose the right platform & application here at WebRTC Conference & Expo IV!

Your Experience This Week Includes:

- Keynotes By: Dialogic, Temasys, Tokbox, Google, Genband, Vidyo, Requestec, Avaya, Oracle
- See and Test WebRTC Products from Top WebRTC Advocates
- 25+ Conference Sessions [Including NEW Tracks: Contact Center, Healthcare]
- Specialized Training:
 - EasyRTC Workshop - Pizza & Wing Night
 - Using WebRTC to Make Money
 - Customer Services: The Killer App for WebRTC
 - Peer to Peer RTC Conference
 - WebRTC for Telecom Professionals

It is a disruptive technology that is altering the way the world communicates – every enterprise is implementing a strategy that integrates WebRTC into their websites, call centers and more. If you have any questions, or ideas for the next event, be sure to connect with a member of the show team.



Contents

Keynote Speakers.....	2	Session Descriptions	12-22
Event Sponsors	4	Exhibitor List/Floor Map.....	26
Agenda-At-A-Glance	6-10	Exhibitor Descriptions	28-34

Tuesday, June 17 - Hall A



Kevin Cook
President & CEO
12:40pm

DiaLogic



Chip Wilcox
CEO
3:10pm

Temasys



Ian Small
CEO
4:10pm

tokbox

Wednesday, June 18 - Hall A



Serge Lachapelle
Product Manager
8:30pm

Google



Brad Bush
EVP & CMO
12:10pm

GENBAND



Alex Eleftheriadis
Chief Scientist & Co-Founder
3:30pm

Vidyo



Ben Weekes
CTO & Co-Founder
4:00pm

requestec

Thursday, June 19 - Hall A



Gary Barnett
Senior VP & GM,
Avaya Collaboration
9:50am

AVAYA



Bernard Aboba
Principal Architect
10:20am

Microsoft



James Steadman
Senior Director,
Service Delivery Solutions
11:45am

ORACLE

Live WebRTC Demos & Reception

**Wednesday
June 18
4:30pm-7:30pm**

**Thursday
June 19
3:00pm-4:45pm**

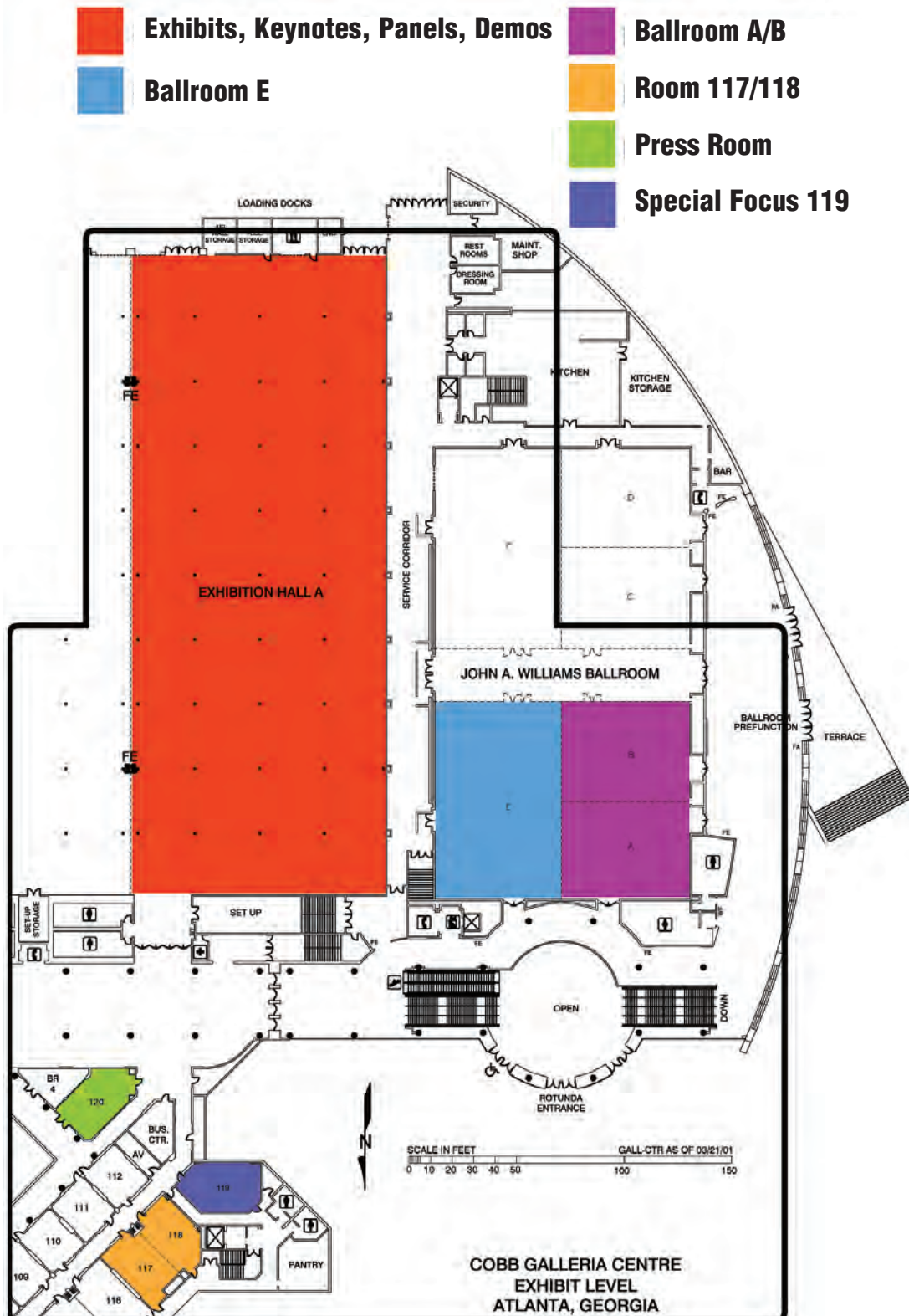
Hall A

FLOORPLAN



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Tuesday, June 17			
TRACK ROOM	Developer	Developer	Business Apps
	Room - Ballroom E	Room - 117/118	Room - Ballroom AB
9:30-10:50am	WebRTC Tutorial and Introductory Training	Deploying WebRTC Successfully The Big Issues	Business Intro to WebRTC Extended Session
10:50-11:00am	Break		
11:00-11:40am	The Big Build Comparison of WebRTC Development Platforms	Delivering Great WebRTC on Mobile Devices	Defining the WebRTC Eco-System
11:40-12:20pm			WebRTC Economics for an Application
12:30-12:40pm	Conference Welcome - Hall A		
12:40-1:40pm	Conference Luncheon - Featuring Keynote by Dialogic - Kevin Cook - Hall A		
1:40-2:20pm	Managing and Scaling WebRTC Video	Data Channel	WebRTC Standards and Regulatory Update for Business
2:20-3:00pm			The Global View
3:00-3:10pm	Break		
3:10-3:40pm	Keynote Presentation by Temasys - Chip Wilcox - Hall A		
3:40-4:10pm	Plenary Panel - Using WebRTC to Make Money - Hall A		
4:10-4:50pm	Keynote Presentation by Tokbox - Ian Small - Hall A		
5:00-7:30pm	Grand Opening Reception in Exhibit Hall - Exhibits Open - Hall A		

* Session subject to change without notice.

Live WebRTC Demos & Reception

**Wednesday
June 18**
4:30pm-7:30pm

**Thursday
June 19**
3:00pm-4:45pm

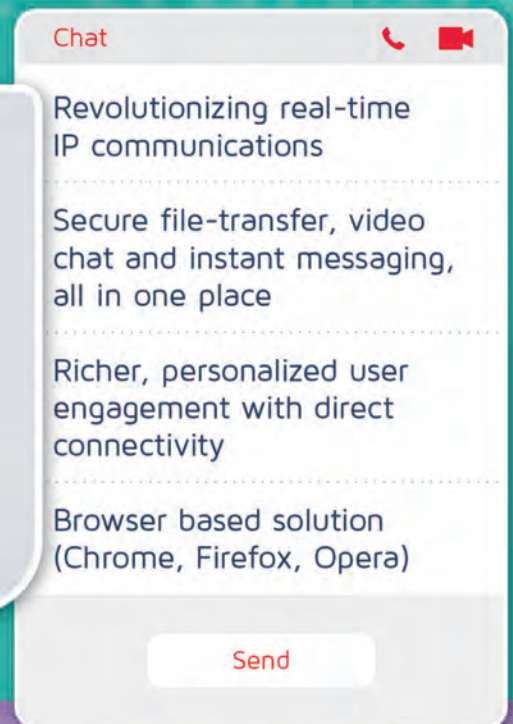
Hall A

Priologic Pizza & Wing Night

priologic

Wednesday, June 18
7:30pm - Ballroom E

acision WebRTC



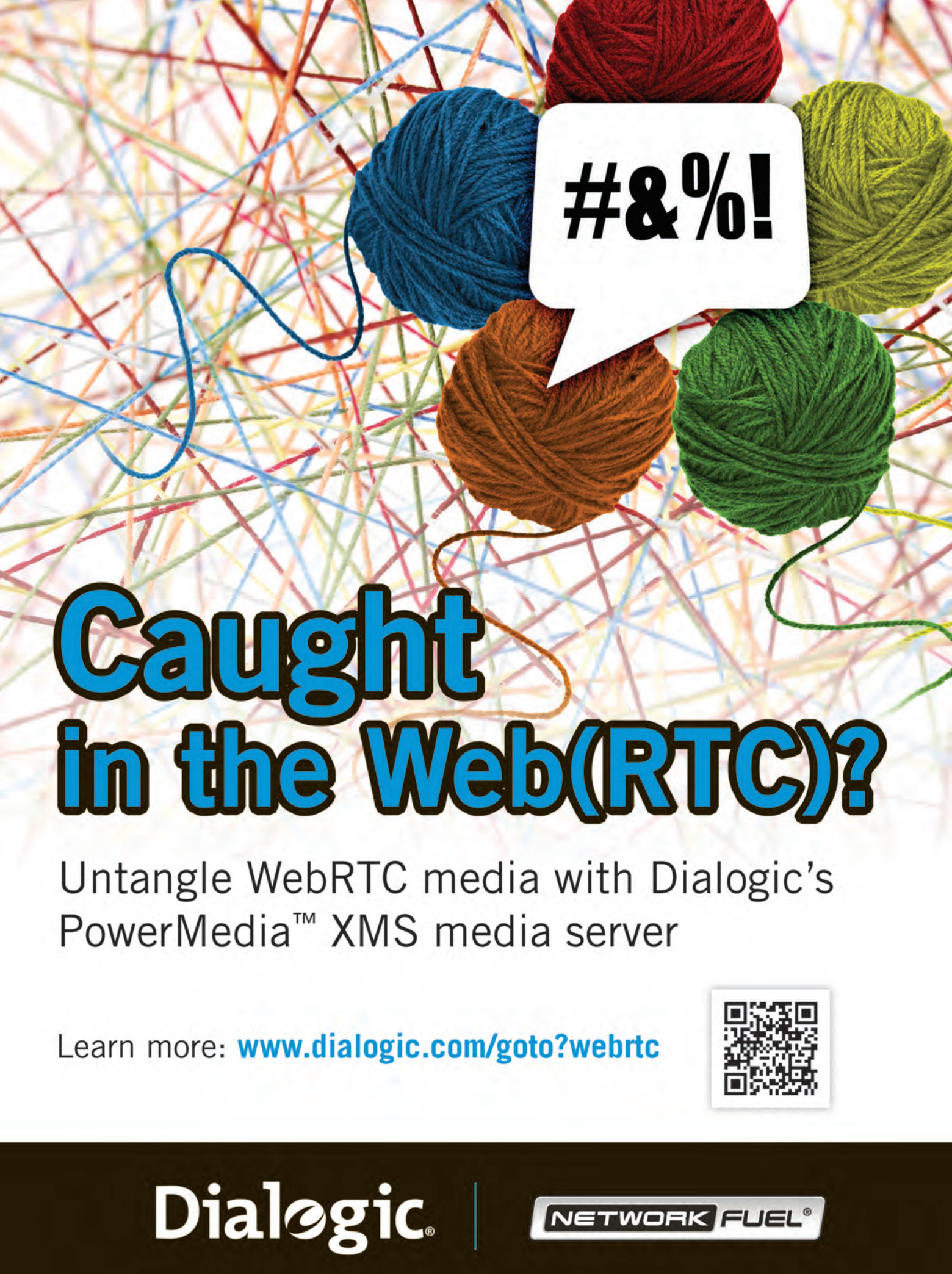
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Wednesday, June 18				
8:00-8:30am	Continental Breakfast - Exhibits Open - Hall A			
8:30-9:10am	Keynote Presentation by Google - Serge Lachapelle - Hall A			
TRACK	Developer	Service Provider	Enterprise	Special Focus
ROOM	Room - Ballroom E	Room - 117/118	Room - Ballroom AB	Room - 119
9:10-9:50am	Standards Update and Directions	Using WebRTC to Deliver Enterprise Services	Will WebRTC Replace Integration of UC into Business Applications?	Healthcare Focus Delivering Remote Patient Care with WebRTC
9:50-10:20am	The Video Conundrum	Security Considerations for Service Providers	Making Your Enterprise Ready for WebRTC	Healthcare Focus - In Room Care Enhancement with WebRTC
10:20-11:30am	Visit the Exhibit Hall - Refreshments Served - Hall A			
11:30-12:10pm	Plenary Panel - Customer Service: The Killer App for WebRTC? - Hall A			
12:10-1:00pm	Conference Luncheon - Featuring Keynote by GENBAND - Brad Bush - Hall A			
1:00-2:00pm	Visit the Exhibit Hall - Coffee & Dessert Served - Hall A			
2:00-2:40pm	Native Apps Versus WebRTC - Options and Performance	Business Model Changes	SIP and WebRTC Working Together?	Healthcare Focus WebRTC Integration with HIMS
2:40-3:20pm	WebRTC Plug-ins for Other Browsers	Service Provider Views	WebRTC in Mixed Environments	Contact Center Focus Webpages to WebRTC
3:20-3:30pm	Visit the Exhibit Hall - Refreshments Served - Hall A			
3:30-4:00pm	Keynote Presentation by Vidyo - Alex Eleftheriadis - Hall A			
4:00-4:30pm	Keynote Presentation by Requestec - Ben Weekes - Hall A			
4:30-7:30pm	WebRTC World Demos & Reception - Hall A			
7:30pm	Priologic Pizza, Wings, Beer & EasyRTC Workshop - Ballroom E			

* Session subject to change without notice.



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	Thursday, June 19			
TRACK	Developer	Service Provider	Enterprise	Special Focus
ROOM	Room - Ballroom E	Room - 117/118	Room - Ballroom AB	Room - 119
8:00-8:30am	Peer to Peer RTC-Conference A Unique Opportunity to Network and Meet the Experts on a Variety of WebRTC Topics	Continental Breakfast - Exhibits Open - Hall A		
8:00-9:40am		Decision Point To IMS or Not: Should Service Providers Integrate WebRTC to IMS or Separate?	Decision Point Contact Center Options and Decisions To Integrate or Do WebRTC Separate? - Ballroom AB	
9:40-9:50am	Visit the Exhibit Hall - Refreshments Served - Hall A			
9:50-10:20am	Keynote Presentation by Avaya - Gary Barnett - Hall A			
10:20-10:50am	Keynote Presentation by Microsoft - Bernard Aboba - Hall A			
10:50-11:45am	WebRTC for Telecom Professionals - What are the 10 Telecom Rules that Do Not Apply? - Hall A			
11:45 -12:45pm	Conference Luncheon - Featuring Keynote by Oracle - James Steadman - Hall A			
12:45-1:30pm	Visit the Exhibit Hall - Coffee & Dessert Served - Hall A			
1:30-2:10pm	Delivering a Great User Experience	WebRTC Deployment Options	New Conferencing Options	Contact Center Focus Feedback to Optimize Self-Service
2:10-2:55pm	Media Services	WebRTC and SDN/NFV and QoS	WebRTC and Enterprise Mobility	Contact Center Focus WebRTC and Agents
3:00pm	WebRTC World Demos - Hall A			
4:50pm	Closing Comments & Giveaway - Must Be Present to Win - Hall A			

* Session subject to change without notice.



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Kickstart your WebRTC implementation today! Learn how at **Table 224**

pizza & wing night!

WEDNESDAY, JUNE 18 @ 7:30pm

RSVP at
Registration

Come socialize, enjoy a beer, meet your new friends, and learn about the fastest and easiest way to implement secure WebRTC!



Developer Track - Tuesday, June 17, 2014

Tuesday, June 17 - Ballroom E

9:30-10:50am

WebRTC Tutorial and Introductory Training

Dan Burnett, Chief Scientist, Voxeo Labs,
Director of Standards, Voxeo, Tropo/Voxeo;
Alan Johnston, Distinguished Engineer and Standards
Leader, Avaya;
Erik Linask (Moderator), Group Editorial Director, TMC

In this session the authors of the new book on how to develop and implement WebRTC using with the new IETF and W3C standards will overview the concepts and structure of WebRTC and how it is defined in the emerging standards. The session will bring everyone up to a clear understanding of WebRTC for the technical discussions going forward. The workshop will include specific examples of how to code and create realtime interactions. The session will be interactive, allowing for open and clear discussion.

Tuesday, June 17 - Room 117/118

9:30-10:50am

Deploying WebRTC Successfully - The Big Issues

Chris Piche, CEO, Eyeball Networks;
John Riordan, CTO, OnSIP;
Carl Ford (Moderator), Partner and Community Developer
Crossfire Media

This workshop will detail all of the elements to actually set up a successful WebRTC deployment and will cover each area and the required pieces to make your WebRTC deployment successful. The session will focus on the key elements of testing.

Tuesday, June 17 - Ballroom E

11:00am-12:20pm

The Big Build Comparison of WebRTC Development Platforms

Rod Apeldorn, Priologic;
Tim Ricciardi, Technology One Group / Tokbox

This workshop will include 4 presentations on detailed implementations based on how WebRTC was built. The options will include from scratch, and using frameworks that are available. The session will contrast the effort and challenges in each approach and conclude with a round table discussion of the differences in approach.

Tuesday, June 17- Room 117/118

11:00am-12:20pm

Delivering Great WebRTC on Mobile Devices

Saraj Mudigonda, Business Development Manager,
Imagination Technologies;
Thomas Sheffler, Application Engineer, Weemo;
Amir Zmora (Moderator), Independent Consultant & AudioCodes
Delivering WebRTC on Mobile devices is a challenge. One option is to go the browser path as more and more Android devices are enabling WebRTC directly in the browser. However, the option of using a WebRTC media engine in a bespoke native client is an option that many developers have leaned to, especially for iOS devices. This workshop will focus on the options and implementation of WebRTC across the mobile environment.

Tuesday, June 17 - Ballroom E

1:40-3:00pm

Managing and Scaling WebRTC Video

Alex Eleftheriadis, Chief Scientist and Co-Founder, Vidyo Inc;
Dan Burnett (Moderator), Chief Scientist, Voxeo Labs,
Director of Standards, Voxeo, Tropo/Voxeo

WebRTC has been primarily deployed as peertopeer services between end points. However, to meet the needs of the market, WebRTC needs to scale to many end points and have the capability to manage codec differences and even devices that can only receive a single video stream. This session will present options for building and managing central video facilities in a large WebRTC deployment.

Tuesday, June 17 - Room 117/118

1:40-3:00pm

Data Channel

The data channel is one of the key features in WebRTC. This sessions will focus on how to se the data channel in new ways and to use it to build connected applications across multiple devices.



Exhibits Open

Tuesday, June 17.....	5:00 – 7:30pm
Wednesday, June 18.....	8:00 – 9:10am.....10:20am – 2:00pm3:20 – 7:30pm
Thursday, June 19.....	8:00 – 8:30am.....9:30am – 1:30pm.....3:00 – 5:00pm

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GENBAND

Developer Track - Wednesday, June 18, 2014

Wednesday, June 18 - Ballroom E
9:10-9:50am

Standards Update & Directions

Alissa Cooper, Cisco;
Adam Roach, Principal Platform Engineer, Mozilla;
Dan Burnett (Moderator), Chief Scientist, Voxeo Labs, Director of Standards, Voxeo, Tropo/Voxeo

In this session key leaders and contributors to the IETF and W3C standards efforts will report on the progress of the standards and how they will come to conclusion and release. This session will be critical in helping understand when to start your WebRTC development and release.

Wednesday, June 18 - Ballroom E
9:50-10:20am

The Video Conundrum

Alex Eleftheriadis, Chief Scientist and co-founder, Vidyo Inc.;
Anatoli Levine (Moderator), Director, Product Management, Americas, Avaya

The future of WebRTC is tied to codecs and how media is managed. Over the past three months a number of new developments in video have come to the fore. Google and Vidyo announced a project to deliver SVC with VP9 and interoperability groups are beginning to focus on WebRTC and video. This session will include leading video experts and standards players, as well as representatives from the UCIF and the IMTC. The panel will focus on where WebRTC video is today and where it is going.

Developer Track - Thursday, June 19, 2014

Thursday, June 19 - Ballroom E
8:00-9:40am

Peer to Peer RTC-Conference

A unique opportunity to network and meet the experts in a variety of WebRTC topics. Topics covered at separate focus tables could include: Getting Started Security Turn Server Data Channel Media Services Testing and Monitoring HTML5 Integration Mobile Google Demo Code Firefox Chrome

Thursday, June 19 - Ballroom E
1:30-2:10pm

Delivering a Great User Experience

AI Cook, Product Marketing, Twilio;
Anton Venema, CTO & Co-Founder, Frozen Mountain;
Amir Zmora (Moderator), Independent Consultant & AudioCodes

In the web world the standards of user experience are always evolving. What is considered best practices changes over time as new techniques evolve. WebRTC introduces a whole new set of elements to the User Experience, from video windows to data channel object synchronization. This session will present current best practices and recommendations to optimize the User Experience of your realtime applications.



Thursday, June 19 - Ballroom E
2:10-2:55pm

Media Services

Ben Weekes, CTO & Co-Founder, Requestec;
Amir Zmora (Moderator), Independent Consultant & AudioCodes
As WebRTC matures, many of the applications and solutions that use it will require more users in a session than can be accommodated with peer to peer video. There are three options for a centralized media server facility: a media mixing MCU, a routing switch, or a hybrid. This session will define the three options, contrast costs and complexity and develop recommendations for classes of applications.

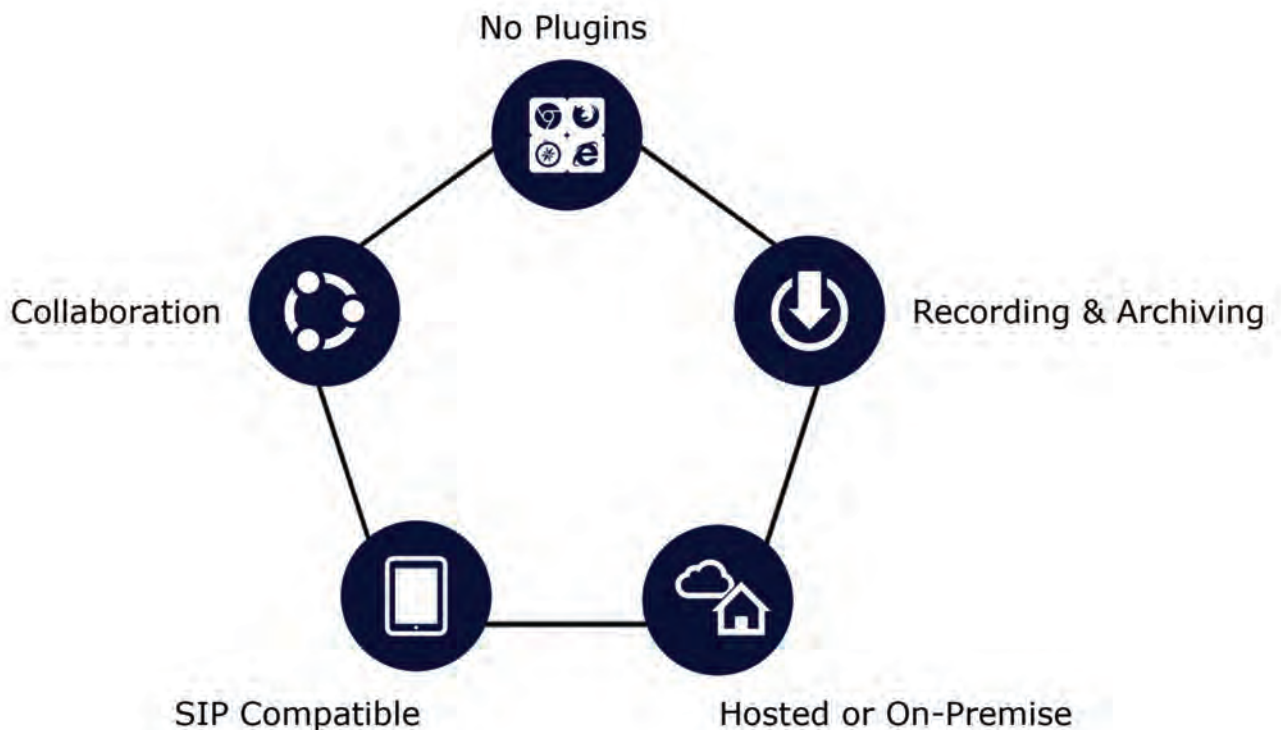




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Business Applications Track - Tuesday, June 17, 2014

Tuesday, June 17 - Ballroom AB
9:30-10:50am

Business Intro to WebRTC - Extended Session

Phil Edholm, President and Principal, PKE Consulting LLC & UC Strategies.com;
Brent Kelly, Vice President and Principal Analyst, Constellation Research

This session is designed to introduce business and nontechnical attendees to what WebRTC is all about. The session will cover the origins of WebRTC, the concepts of a Media Engine and its purpose in a IP edge device and the concepts of triangles and trapezoids. The session will contrast WebRTC and SIP as well as discuss how WebRTC can be used to generate new capabilities. The session will include an analysis of the potential impact of WebRTC contrasted with the impact of the browser/web and VoIP.

Tuesday, June 17 - Ballroom AB
11:00-11:40am

WebRTC Eco-System

Badri Rajasekar, TokBox

WebRTC is a technology and has a complete ecosystem of implementers, providers, users, and developers. This session will overview the WebRTC ecosystem and how the different components interact. It is critical for anyone planning a major WebRTC deployment and looking to understand how to engage to deliver a complete solution.

Tuesday, June 17 - Ballroom AB
11:40am-12:20pm

WebRTC Economics for an Application

Val Matula, Senior Director, Multimedia Technologies, Avaya;
Alan Percy, Senior Director of Strategic Marketing for North America, AudioCodes;
Brent Kelly (Moderator), Vice President and Principal Analyst Constellation Research

This session will examine the economics of WebRTC and how it applies to applications, both communications focused and general. The session will cover development and deployment costs, including elements such as TURN server, media servers as well as app servers. The session goals is to provide a set of data that can be used to financially model WebRTC deployments.

Tuesday, June 17 - Ballroom AB
1:40-2:20pm

WebRTC Standards and Regulatory Update for Business

Martha Buyer, Principal, Law Offices of Martha Buyer;
Tim Panton, Director, Westhawk Ltd

This session will start with a quick nontechnical update on the standards and then focus on regulatory issues. With many recent regulatory moves around VoIP, government recording, and other areas, understanding the regulatory response to WebRTC is a critical part of the overall strategy, especially for Service Providers. The session will cover how nonFCC regulation like CALEA and 911/Emergency Services may impact WebRTC deployment models.

Tuesday, June 17 - Ballroom AB
2:20-3:00pm

The Global View

David Alozie, Retail Geek;

Etisalat Nigeria;

Chip Wilcox, Tamasys

Brent Kelly (Moderator), Vice President and Principal Analyst, Constellation Research

While much activity in WebRTC in North America and Europe, the reality is that WebRTC adoption may explode in other areas of the world as well. This session will have experts from Asia and Africa discussing the impact and adoption of WebRTC in those geographies and the potential impact on business, investment, and societies.



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IS TO KNOW SOMETHING
NOBODY ELSE KNOWS ”**

— Aristotle Onassis



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Service Provider Track - Wednesday, June 18, 2014

Wednesday, June 18 - Room 117/118

9:10-9:50am

Using WebRTC to Deliver Enterprise Services

Micaela Giuhath, Sr. Director Solutions, BroadSoft;

John Riordan, CTO, OnSIP;

Chip Wilcox, Temasys;

Rich Tehrani (Moderator), CEO, TMC

This session will focus on how WebRTC can be used to deliver new and advanced services to enterprises, ranging from Unified Communications to Contact Center. It will also look at how Service Providers are and can use WebRTC in their own operations.

Wednesday, June 18 - Room 117/118

9:50-10:20am

Security Considerations for Service Providers

Glen Gerhard, VP Product Management, Sansay

While WebRTC has builtin security, there are a number of additional concerns for Service Providers. This session will look at other security concerns, from identity to spam and spoofing. The session will discuss options for securing a WebRTC deployment and managing the security threat environment, including media servers and other resources.

Wednesday, June 18 - Room 117/118

2:00-2:40pm

Business Model Changes

Brad Bush, CMO, Genband;

Jack Kozik, Alcatel Lucent;

Dean Bubley (Moderator), Founder, Disruptive Analysis

WebRTC brings new technologies, economics, and implementations to the Service Provider marketplace. With capabilities such as PeerCDN and the data channel, there are new opportunities for monetization. This panel will focus on the impact of WebRTC on the Service Provider Business Model and the [potential of new economics driving new revenue opportunities.

Wednesday, June 18 - Room 117/118

2:40-3:20pm

Service Provider Views

Dag-Inge Aas, Software Engineer, Telenor Digital - appear.in;

Ian Small, CEO, Communications Services, Telefonica;

Dean Bubley (Moderator), Founder, Disruptive Analysis

This session will feature Telenor and other service providers talking to how they see using WebRTC in their offerings to enhance services and revenue.

Service Provider Track - Thursday, June 19, 2014

Thursday, June 19 - Room 117/118

8:30-9:40am

To IMS or Not - Should Service Providers Integrate WebRTC to IMS or Separate

Russ Coffin, Strategic Architect, Huawei Technologies USA;

Graham Holt, Executive Vice President Sales and Marketing, Daitan Group;

Douglas Tait, Global Director of Marketing for Telecommunications, Oracle;

Michael Trank, Product Manager, Apex Voice;

Dean Bubley (Moderator), Founder, Disruptive Analysis

Most Service Providers have made a major investment in IMS and how they manage and operate their networks, especially for Mobile.

With the advent of Chrome 29 and next generation Mozilla mobile apps, WebRTC will be available on mobile as well as terrestrial devices. This session will contrast integrating WebRTC into the existing IMS architecture or developing a WebRTC solution and product set separately. The focus will be on the cost and values of IMS integration versus the simplicity and speed of a separate solution set. The session will include two presentations of the values and a lively debate.

Thursday, June 19 - Room 117/118

1:30-2:10pm

WebRTC Deployment Options

David Buechner, Vice President, Eyeball Networks;

John Parr, Managing Director, Acision;

Carl Ford (Moderator), Partner and Community Developer, Crossfire Media

Developing a deployment strategy for how to use and deploy WebRTC is a key challenge faced by many Service Providers. From adjunct to existing services, extending value for users to new devices, or offering services to new users covered by other access networks, the options are many. This session will focus on how to deploy WebRTC and how to deliver services.

Thursday, June 19 - Room 117/118

2:10-2:55pm

WebRTC and SDN/NFV and QoS

Graham Holt, Executive Vice President Sales and Marketing, Daitan Group;

Carl Ford (Moderator), Partner and Community Developer, Crossfire Media

How will WebRTC integrate with SDN/NFV and other changes in the network. Will the networks be capable of delivering the QoS that WebRTC applications require. The session will review the changes in the network as they relate to WebRTC and mechanisms for integrating WebRTC into the network infrastructure.



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Enterprise Track - Wednesday, June 18, 2014

Wednesday, June 18 - Ballroom AB
9:10-9:50am

Will WebRTC Replace Integration of UC into Business Applications?

Thomas Cottreau, CEO, Weemo;
Michal Raz, Vice President of Business Development, Vidyo
Amir Zmora (Moderator), Independent Consultant & AudioCodes
Since the beginning of the Unified Communications business in 2005, there has been a discussion of the two sides of UC, UCC: integration of personal communications tools, presence, IM, and video for Knowledge Workers, and UCB: the integration of communications into formal business process for Information and Service Workers. While there is great agreement that the benefits of UCB are huge, the difficulties of implementation have been huge as well. Now the major UC vendors are touting applications integration, but WebRTC promises the potential of communications built into the app and not integrated with the UC platforms. This panel will debate this and discuss options and paths forward.

Wednesday, June 18 - Ballroom AB
9:50-10:20am

Making Your Enterprise Ready for WebRTC

Karl Stahl, CEO, Ingate Systems AB;
Amir Zmora (Moderator), Independent Consultant & AudioCodes
Deploying WebRTC requires both a technical implementation, but also a number of other factors to assure success. This session will discuss the areas you need to focus on, from devices that will be used, through web security to organizational integration and management. The session will let you understand what you need to do to make your WebRTC deployment succeed.

Wednesday, June 18 - Ballroom AB
2:00-2:40pm

SIP & WebRTC - Working Together?

Joel Ezell, Sr. Architect, Research & Development, Avaya;
Alan Percy, Senior Director of Strategic Marketing for North America, AudioCodes;
Carl Ford (Moderator), Partner and Community Developer, Crossfire Media
While WebRTC is powerful and has huge open opportunities on the Internet, most enterprises are just now deploying SIP as a way to normalize and reduce costs in their communications infrastructure. This session will focus on the challenges and methods of integrating WebRTC into SIP, either through a vendor solution or the open options available. It will discuss the relative benefits of direct media interoperation versus using a media server of translation as well as the emerging support from vendors for WebRTC components like VP8. This session is essential to mapping out your integrations strategy between WebRTC and SIP.

Wednesday, June 18 - Ballroom AB
2:40-3:20pm

WebRTC in Mixed Environments

Brad Bush, CMO, Genband;
Al Cook, Product Marketing, Twilio;
Carl Ford (Moderator), Partner and Community Developer, Crossfire Media
With the potential that many early rollouts of WebRTC cannot be 100% WebRTC due to the lack of finalized codecs and the non-support in both Internet Explorer and Safari, there is a requirement for network-based services to integrate the mixed environments. This session will examine how network elements such as SBCs, gateways etc can enhance the deployment options and provide the integration necessary.

EnterpriseTrack - Thursday, June 19, 2014

Thursday, June 19 - Ballroom AB
8:30-9:40am

Contact Center Options & Decisions - To Integrate or Do WebRTC Separately

Val Matula, Senior Director, Multimedia Technologies, Avaya;
Chris Vitek (Moderator), Principal, WebRTC Strategies
Almost everyone agrees that Customer Interaction is one of the first places WebRTC may have a major impact, in fact, in the WebRTC World Outlook 2014, Contact Center was by far the largest identified target market, both as an opportunity and for investment. This session will discuss the options going forward, of integrating together your existing web site and Contact Center of using new technologies and applications for a more seamless approach. The session will include options and a lively debate.

Thursday, June 19 - Ballroom AB
1:30-2:10pm

New Conferencing Options

Arnaud Budkiewicz, Co-Founder & CEO, Bistri;
Brent Kelly, Vice President and Principal Analyst, Constellation Research;
Will Mitchell, Lead Developer, OnSip;
Gregor Zebic, CMO, Browssetel
WebRTC has already opened up a number of new conferencing options, from audio only, to video and full video/web conferencing. This session will examine the state of the conferencing industry with WebRTC, including factors such as legacy integrations, centralized facilities, scale, and other factors.

Thursday, June 19 - Ballroom AB
2:10-2:55pm

WebRTC & Enterprise Mobility

Brent Kelly, Vice President and Principal Analyst, Constellation Research;
Doug Pelton, CEO, Priologic
With WebRTC, enterprises can begin on a new path to mobility. As there is no longer a need to download a client for the communications media engine, the potential of having devices operate totally clientless and dataless may dramatically simplify the challenges of BYOD and mobility. This session will examine these trends and discuss how mobility will be impacted by WebRTC.





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Special Focus - Wednesday, June 18, 2014

Wednesday, June 18 - Room 119

9:10-9:50am

Healthcare Focus

Delivering Remote Patient Care with WebRTC

Jim Donovan, VP, Product Management, Oracle - Acme Packet;
Bruce Marler, Director of Sales Engineering, CafeX Communications;

Brent Kelly (Moderator), Vice President and Principal Analyst Constellation Research

WebRTC has been positioned as an ideal technology to enable remote patient monitoring. The capability to include video and even realtime data in a care giver to patient session increases the breadth of these capabilities. The ability to have the solution operate without complex downloads enables services for a broader group of patients. This session will discuss applications of WebRTC to remote care from a clinic location, either followup or diagnostic. This session will focus on the technical issues and implementations of these solutions.

Wednesday, June 18 - Room 119

9:50-10:20am

Healthcare Focus

In Room Care Enhancement with WebRTC

Soufiane Hourri, VP of Product, Weemo;

Marek Zwiefka-Sibley, Requestec;

Brent Kelly (Moderator), Vice President and Principal Analyst Constellation Research

WebRTC can be used to enhance the inroom experience, as well as to enable video systems that are being installed to do more than just monitor patients. How can inroom care be enhanced with WebRTC based solutions, from either remote health care participants, or family, etc. How can remote specialists be included in both room, ER, and operating events? How should this be implemented and used in a facility. This session will explore how WebRTC can enable these investments to enhance overall service and outcomes.

Wednesday, June 18 - Room 119

2:00-2:40pm

Healthcare Focus

WebRTC Integration with HIMS

Arnaud Budkiewicz, Co-Founder & CEO, Bistri;

Dick Govatski, CEO, Net Medical;

Michal Raz, Vice President of Business Development, Vido;

Tim Panton (Moderator), Director, Westhawk Ltd

A key to using WebRTC in the healthcare industry is integrating it into existing Healthcare Information and Management Systems HIMS. Many of the existing HIMS systems have some form of communications control, and linking that with WebRTC will be critical for the integration of WebRTC into the health care process. This session will examine the state of readiness for that integration and how it can be accomplished.

Wednesday, June 18 - Room 119

2:40-3:20pm

Contact Center Focus

Webpages to WebRTC

Hugh Goldstein, VP, Strategic Alliances, Voxbone;

Kris Hopkins, CafeX Communications;

Chris Vitek (Moderator), Principal, WebRTC Strategies

One of the key advantages to WebRTC in the overall customer experience is the potential for seamless and contextual transitions from the selfservice website to the agent supported interaction. This session will focus on how to map from the web site to the agent/human. What are the tools that area available? How can we take into account the web experience as a precursor to the human experience? How do we use the array of context from the web visit to optimize the agent interaction?

Special Focus - Thursday, June 19, 2014

Thursday, June 19- Room 119

1:30-2:10pm

Contact Center Focus

Using WebRTC and Feedback to Optimize Self-Service

One major potential of WebRTC is to provide a framework for optimizing the selfservice experience on the web site based on the actions taken by agents. If all exits from the web site to an agent represent problems that customers would have preferred to solve using web selfservice, the information about those can transform the web site. This session will focus on how to use the information from the CC to optimize the selfservice web site.

Thursday, June 19 - Room 119

2:10-2:55pm

Contact Center Focus

WebRTC and Agents

Vladimir Beloborodov, UC CTO, MERA NN;

Val Matula, Senior Director, Multimedia Technologies, Avaya;

Michal Raz, Vice President of Business Development, Vido

WebRTC can impact the agent in two major ways; video and remote agent. This session will focus on the agent in the WebRTC enabled Contact Center. Both the use of WebRTC for remote agent enablement and the issues with video services will be discussed and the issues that must be considered will be covered.

Exhibits Open

Tuesday, June 17.....5:00 – 7:30pm

Wednesday, June 18.....8:00 – 9:10am.....10:20am – 2:00pm3:20 – 7:30pm

Thursday, June 19.....8:00 – 8:30am.....9:30am – 1:30pm.....3:00 – 5:00pm

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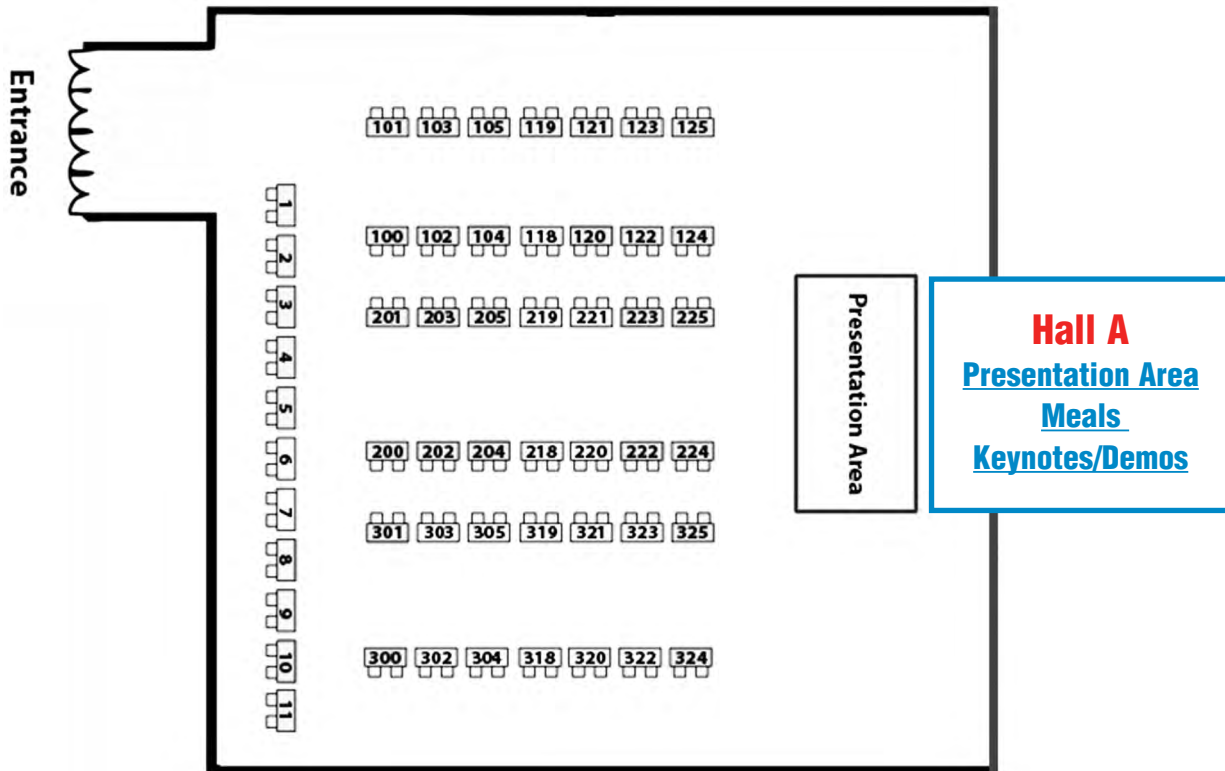
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Exhibitor List

Company	Booth	Company	Booth
Acision.....	8,9	Ice Warp	119
AN Software	105	Imagination	103
APEX Voice Communications.....	203	Ingate Systems	101
AudioCodes.....	225	Mera	121
Avaya	1,2	OnSIP	301
Bistri.....	123	OpenClove	118
Browsetel Ltd.	222	Oracle.....	10,11
CaféX Communications	104	Priologic	224
Content Boost.....	7	Requestec	324
Daitan Group	205	Sansay, Inc.....	300
Dialogic	201	Talkpoint	Sponsor
Digium, Inc.	325	Temasys Communications Ltd	4,5
Eyeball Networks	122	TokBox.....	100,102
Frozen Mountain	223	Twilio	202,204
Genband	3	Videotion	220
Google.....	124	Vidyo	120
Huawei	125	Weemo	200



Session Border Controllers (SBC) for **WebRTC** Application Interoperability

Enterprises



Contact Centers



Service Providers



WebRTC

Visit us at booth #225



Acision (Booth: 8, 9)

P: 07967813429

www.acision.com

Acision connects the world by powering relevant, seamless mobile services that interoperate across all IP platforms and enrich the user experience, creating value and new communication opportunities for carriers, enterprises and consumers across the world. Comprising an open and standardised media profile, Acision WebRTC is a game changing technology delivering secure, carrier-grade, peer-to-peer communications. Enterprises can build powerful web-based and mobile communication applications through the Acision WebRTC SDK with its rich set of APIs, all without the need for any additional infrastructure.

AN Software (Booth: 105)

P/F: +972544982886 / +97288603414

www.an-sw.com

AN Software helps good companies to develop great software. Among AN Software customers are world leading technology companies serving more than a billion users around the world. AN Software helps its customers to integrate WEBRTC voice and video into mobile platform. AN Software solutions run as part of our client products in hundreds of millions of mobile devices.

APEX Communications (Booth: 203)

P/F: 818-379-8400 / 818-379-8410

www.apexcomm.com

APEX is a global provider of Service Delivery Platforms (SDPs) and SIP Application Servers for Voice, Video, SMS and USSD Value-Added Services, which support WebRTC in solutions like Video Conferencing, Hosted ACD, and Self-Service Optimization, with Instant Messaging/Chat. With over 15,000 installations across 100 countries since 1989, APEX continues to be a leader in delivering the latest solutions for mission-critical and revenue-generating enhanced services to Network Operators, Value Added Service Providers, Contact Centers, and Enterprises.

AudioCodes (Booth: 225)

P/F: 888-586-4743 / 732-469-2298

www.audiocodes.com

AudioCodes, a leading manufacturer of voice networking equipment and software solutions enables WebRTC implementations with: - WebRTC-enabled IP Phones with HD Opus support. - Session Border Controllers for WebRTC to SIP integrations - Call recording solutions for WebRTC applications To learn more, visit: www.audiocodes.com/webrtc

Avaya (Booth: 1, 2)

P/F: 800-GO-AVAYA / 972-745-5041

www.avaya.com

Avaya enables businesses to achieve superior results by designing, building, and managing their communications infrastructure. For over one million businesses, including over 90 percent of the FORTUNE 500®, Avaya solutions help businesses enhance value and create competitive advantage by allowing people to be more productive and create more intelligent processes. For businesses large and small, Avaya is a world leader in secure, reliable IP telephony systems, communications applications, and full life-cycle services. Driving the convergence of embedded voice and data communications with business applications, Avaya is distinguished by its combination of comprehensive world-class products and services. For more information, visit www.avaya.com.

Bistri (Booth: 123)

P: 415-283-6899

<https://bistri.com>

Bistri provides a "1-click" video calling app aiming to replace Skype as the next-generation for video communication.

Browsetel Ltd. (Booth: 222)

P/F: +442035148811 / +442036080189

www.browsetel.com

BrowseTel is a cloud service provider offering a platform for real-time multi-channel communication within the Internet browser. A complete telecommunication suite based on WebRTC technology with HTML5 Web Phone with all the functionalities of a desk phone or mobile phone where users can interact online (audio/video/data) as well as on a voice telephone network. BrowseTel.com is also offering API interfaces for integration of existing solutions or integration of 3rd party applications.

CaféX Communications (Booth: 104)

P/F: 646-351-0454 / 646-351-0700

www.cafex.com

CaféX is a leading provider of real-time customer engagement solutions. By providing software solutions for WebRTC and mobile B2C collaboration, CaféX seamlessly connect customers to businesses from within mobile and web applications.

Content Boost (Booth: 7)

P/F: 203-852-6800 / 203-295-3721

www.contentboost.com

Content Boost, a one-stop-shop content strategy vendor, allows you to leverage an arsenal of experienced editorial, marketing, social media and design talent to produce high-quality, engaging, customized content for your company. Designed to amplify your sales, Content Boost is TMC's latest innovation. The Content Boost team assumes responsibility for conceptualizing and originating custom copy, marketing strategy, social media tactics, and multimedia campaigns for our clients.

Daitan Group (Booth: 205)

P/F: 925-475-8732 / 925-475-8739

[/www.daitangroup.com](http://www.daitangroup.com)

Daitan Group is the consulting and development partner-of-choice for communication solution providers. Over the past 30 years we've helped our clients develop and deploy their telecom services in over 50 countries. Today, our engineering team provides WebRTC expertise for contact centers, video/web conferencing, and cool new mobile applications. www.daitangroup.com

Dialogic (Booth: 201)

P/F: 800-755-4444 / 800-755-4444

www.dialogic.com

Dialogic, the Network Fuel™ company, inspires the world's leading service providers and application developers to elevate the performance of media-rich communications across the most advanced networks. Dialogic® PowerMedia™ energizes WebRTC application development and cloud-ready service deployment with multi-party conferencing, media interworking, recording, rich media mixing, and person-to-machine capabilities. Forty-eight of the world's top 50 mobile operators and nearly 3,000 application developers rely on Dialogic to redefine the possible and exceed user expectations.



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For the Carrier



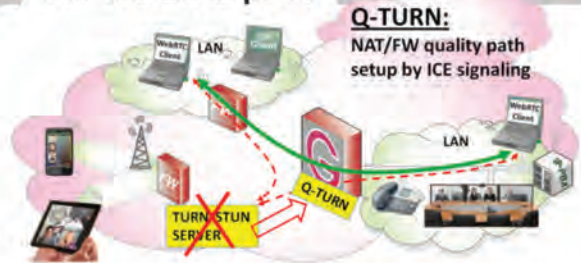
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Digium, Inc (Booth: 325)

P: 256-428-6000

www.digium.com

Digium®, Inc. provides Asterisk® software, telephony hardware, and hosted and on-premises Switchvox business phone systems that deliver enterprise-class Unified Communications (UC) and UC as a Service (UCaaS) solutions at an affordable price. Digium is the creator, primary developer and sponsor of the Asterisk project; the world's most widely used open source communications software. Asterisk turns an ordinary computer into a feature-rich communications server. A community of more than 80,000 developers and users worldwide uses Asterisk to create VoIP communication solutions in more than 170 countries. Since 1999, Digium has empowered developers to create innovative communications solutions based on open standards and open source software, providing an alternative to proprietary phone systems. Digium Switchvox Cloud and other cloud-based products and services are offered through Digium Cloud Services, LLC, a wholly owned subsidiary of Digium, Inc. Digium's business communications products are sold through a worldwide network of reseller partners. More information is available at <http://www.digium.com> and <http://www.asterisk.org>.

Eyeball Networks (Booth: 122)

P/F: 604-921-5993 / 604-921-5909

www.eyeball.com

Eyeball pioneered the STUN/TURN/ICE NAT traversal technologies adopted by communications standards including SIP, XMPP, IMS, PacketCable, Microsoft Lync, and WebRTC. Today Eyeball's AnyFirewall™ Technology is patented, foundational to STUN/TURN/ICE, and guarantees connectivity for the world's largest developers of semiconductors, smartphones, tablets, and VoIP and video telephony systems. Eyeball provides free STUN/TURN services for WebRTC applications at AnyFirewall.com AnyFirewall Technology is embedded in products used by 98% of the Fortune 500 and 100% of the Fortune 100.

Frozen Mountain Software (Booth: 223)

P: 888-FRZN-MTN

www.frozenmountain.com

Frozen Mountain Software is the industry leader in creating powerful commercial software libraries for real-time applications. Whether you want to push data from your server to web browsers, create an elegant REST API, build a fast-action multiplayer game, or add video chat to your application, we have libraries that will get you there quickly and painlessly.

GENBAND (Booth: 3)

P/F: 972-521-5800 / 972-521-5837

www.genband.com

GENBAND is a leading developer of multimedia and cloud communications solutions with service provider and enterprise customers in over 80 countries. The company has established itself as a market-leader in WebRTC technology with its award-winning SPiDR™ WebRTC Gateway solution, which seamlessly bridges the gap between communications networks and the Web. For more information visit <http://www.genband.com/solutions/industry-innovation/webrtc>.

Google (Booth: 124)

P/F: 650-253-0000 / 650-253-0001

www.google.com

Huawei Technologies USA (Booth: 125)

P/F: 214-919-6000 / 214-919-6000

www.huawei.com

Huawei is a leading global information and communications technology (ICT) solutions provider. Through our dedication to customer-centric innovation and strong partnerships, we have established end-to-end advantages in telecom networks, devices and cloud computing. We are committed to creating maximum value for telecom operators, enterprises and consumers by providing competitive solutions and services. Our products and solutions have been deployed in over 140 countries, serving more than one-third of the world's population.

IceWarp (Booth: 119)

P/F: 571-481-4611 / 703-563-9269

www.icewarp.com

IceWarp is a vendor of highly integrated solution IceWarp Server, including Mail server with dual Anti-Spam & Anti-Virus protection and available add on options including the GroupWare server, Instant Messaging server, Text Messaging server, a unified WebClient interface, full mobile device synchronization via Exchange ActiveSync protocol and much more.

Imagination Technologies (Booth: 103)

P/F: 408-530-5000 / 408-530-5150

<http://imgtec.com>

Imagination's HelloSoft range of licensable IP includes comprehensive software SDKs delivering everything needed to implement high quality, power-efficient Voice over IP and LTE (VoIP, VoLTE), Voice plus Video over IP (V.VoIP), rich communications suite (RCS 5.0) and enhanced WebRTC solutions for multi-mode wireless and wireline devices. HelloSoft IP products and technologies are 'carrier-grade' cross-platform solutions, already deployed in millions of devices. Built on years of engineering experience to ensure that they are ready for the most extensive 3G, 4G and enterprise network deployments, all SDKs and platforms are available for Android, iPhone, Windows and Linux environments. Imagination Technologies - a global leader in multimedia, processor and communication technologies - creates and licenses market-leading processor solutions including graphics, video, vision, CPU & embedded processing, multi-standard communications, cross-platform WebRTC, V.VoIP, & VoLTE and cloud connectivity. These silicon and software intellectual property (IP) solutions for systems-on-chip (SoC) are complemented by an extensive portfolio of software, tools and ecosystems.

Ingate Systems (Booth: 101)

P/F: 603-883-6569 / 413-460-0414

www.ingate.com

Ingate Systems develops secure Enterprise Session Border Controllers (E-SBCs) and firewalls that allow the best of SIP- and WebRTC-based quality multimedia real-time communication and SIP trunking globally beyond the protected enterprise network. Ingate Systems develops the WebRTC & SIP PBX Companion, allowing PBX and UC solution vendors to integrate all WebRTC features and benefits into their enterprise solutions. Ingate also develops the Q-TURN technology that will be integrated in its E-SBCs giving WebRTC, or any real-time communication using the ICE standard for NAT/firewall traversal, secure access and quality between the enterprise LAN and the World Wide Web, even over data-crowded Internet accesses.



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Mera Software Services (Booth: 121)

P: 214-471-0117

www.merasws.com

MERA is a leading provider of outsourced software development services for the ICT industry. Since the inception in 1989 the company has turned into a trusted R&D partner to the top equipment manufacturers and solution vendors making the robust world of communications more seamless, reliable and predictable.

OnSIP (Booth: 301)

P: 212-933-9190

www.onsip.com

Now supporting over 25,000 customers, The OnSIP Network is a signaling Platform as a Service for companies to rapidly and affordably develop and maintain WebRTC applications. Developers benefit from simple and intuitive APIs, including open sourced SIP.js, to easily add signaling to their WebRTC applications without the burden of building their own stack. For more information, visit www.onsip.com/webrtc

OpenClove (Booth: 118)

P/F: 469-713-6736 / 469-713-6736

www.openclove.com

OpenClove provides Communications for the Internet of Apps. We enable real-time engagement across education, healthcare, e-Commerce, financial, human resources, social, gaming, content and collaboration applications. Our customers are enterprises, mobile app developers and web service providers, who use our Secure Communications as a Service (SCaaS) to deliver real-time video, voice conversations and data interactivity to their users within their own apps, across any device platforms - WebRTC, iOS and Android.

Oracle Communications (Booth: 10, 11)

P/F: 800-417-0059 / 781-744-0001

www.oracle.com

Oracle Communications solutions span the communications industry landscape – from cross-channel customer experience and business and operational support systems, to network service and session delivery and control solutions – enabling service providers and enterprises to deliver and monetize innovative digital lifestyle services, build strong customer relationships, and streamline operations. For more information, visit <http://www.oracle.com/communications>.

Priologic Software (Booth: 224)

P/F: 250-940-9080 / 250-940-9086

<http://easyrtc.com/>

Priologic is the developer of award winning WebRTC products and EasyRTC a free, open source WebRTC toolkit that makes building basic Enterprise WebRTC apps quicker and easier. EasyRTC Enterprise a licensed WebRTC toolkit that extends EasyRTC with native SDKs for iOS and Android and other platforms.

Requestec Saypage (Booth: 324)

P/F: +44 (0) 8707 505 896 / +44 (0) 8707 505 896

www.requestec.com/

Through our Saypage Communication & Collaboration Platform we provide developers with SDKs and APIs to connect users across all web browsers and iOS and Android devices. Supporting multi-party conferencing and presenting, together with collaboration tools such as whiteboard and screen-sharing,

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Sansay, Inc. (Booth: 300)

P/F: 760-815-4016 / 858-550-2044

www.sansay.com/

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TalkPoint (Booth: Sponsor)

P: 212-909-2900

www.talkpointcommunications.com

TalkPoint offers a robust cloud-based, self-service webcasting technology for companies to improve communications with their internal and external constituencies. The Convey™ webcasting platform is tailored to the needs of IT professionals, yet designed for business users to self-manage a 50-person video-enabled meeting as efficiently as a webcast to 10,000 attendees. Seamlessly leverage existing enterprise assets such as videoconferencing units, encoders and webcams to deliver professional-quality webcasts to viewers on desktop and mobile devices – all through a single URL within the browser. No downloads. No plugins. No apps. For more information, visit

Temasys Communications Pte Ltd (Booth: 4, 5)

P/F: 656-224-2558 / 656-224-2558

www.temasys.com

Temasys is one of the world's leading innovators in WebRTC. Temasys provides the infrastructure and developer tools to enable the new wave of innovation in web and mobile communications based on WebRTC. Our customers use our solutions to improve service delivery, to increase business collaboration and decision-making velocity, and to make online social interaction more accessible and fun.

TokBox (Booth: 100, 102)

P/F: 925-989-4791 / 415-284-4610

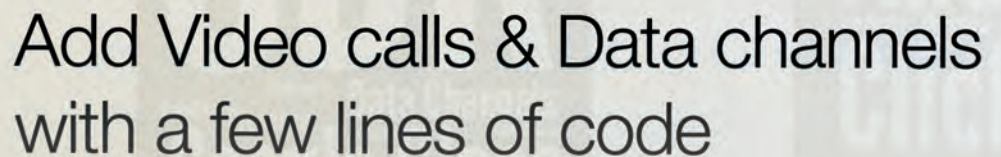
<http://tokbox.com>

TokBox, a Telefónica company, develops and operates OpenTok, a global platform that enables you to quickly and easily integrate real-time communications into your website and mobile apps. Whether you're developing an app for one-to-one calls or for complex large-scale broadcasts, OpenTok's comprehensive API helps you deliver a world class communications experience powered by WebRTC. OpenTok offers industry leading features including multi-party calling, call archiving, mobile support, Intelligent Quality Controls and more. OpenTok has been used by companies including Major League Baseball, Diet Coke, Ford, Doritos, Bridgestone Golf and Double Robotics. TokBox is headquartered in San Francisco, CA.



Open WebRTC Platform

iOS



Gartner. 2013
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Twilio (Booth: 202, 204)

P: 415-682-6419

www.twilio.com

Twilio provides a software and cloud-based communications platform that enables developers and businesses to build the communication solution that meets their specific needs. Whether integrating voice, messaging or VoIP capabilities into a web or mobile app or building a complete call center solution, Twilio removes traditional obstacles to creating effective communications experiences. By virtualizing the entire infrastructure needed for communications in a cloud-based, global environment, communications solutions are simple to build, scalable and easily adapted to changing business needs. By leveraging Twilio, businesses are better able to connect with their customers in the way their customers want to communicate.

Videon (Booth: 220)

P/F: 425-241-5006 / 425-241-5006

www.videon.com

Real Time Communication and Sharing as a Service (RTC&SaaS) Provider

Vidyo, Inc. (Booth: 120)

P/F: 866-998-4396 / 201-490-5340

www.vidyo.com

VidyoWorks™ is the WebRTC media platform that enables companies to integrate high quality video into their applications and achieve scale, quality, and custom/integrated solutions. The VidyoRouter™ leverages Scalable Video Coding (SVC) which enables better error resilience and allows an optimized experience for multi-user video calls. Web developers using the VidyoRouter in their WebRTC implementation will have access to scalable, high quality video without the need for a separate client or plug-in installation. An inherent benefit of the WebRTC architecture and Vidyo's collaboration with Google, is that developers will be able to use the scalable coding extensions of VP9 without significant changes to their client code and, additionally, benefit from the power of the VidyoRouter to drive high quality novel applications at large enterprise and service provider scale. Learn more at www.vidyo.com, on the Blog or follow Vidyo on Twitter @vidyo and on Facebook.

Weemo (Booth: 200)

P: 415-735-3422

www.weemo.com

Weemo is a video cloud platform that easily enables real-time video into applications serving different contexts such as customer service, healthcare or enterprise collaboration. Weemo provides a development framework composed of APIs and mobile SDKs as well as ready-to-use add-ons. Weemo relies on WebRTC as well as other best of breed technology to make real-time video available on every browser and device at a global scale. For more information go to: <http://weemo.com/> or follow the company @Weemo

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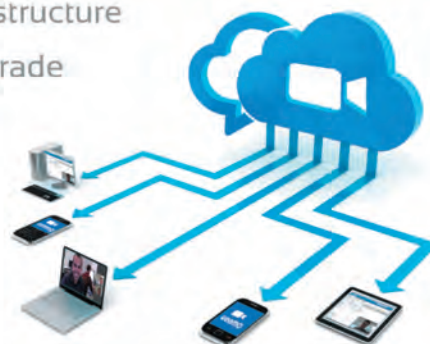
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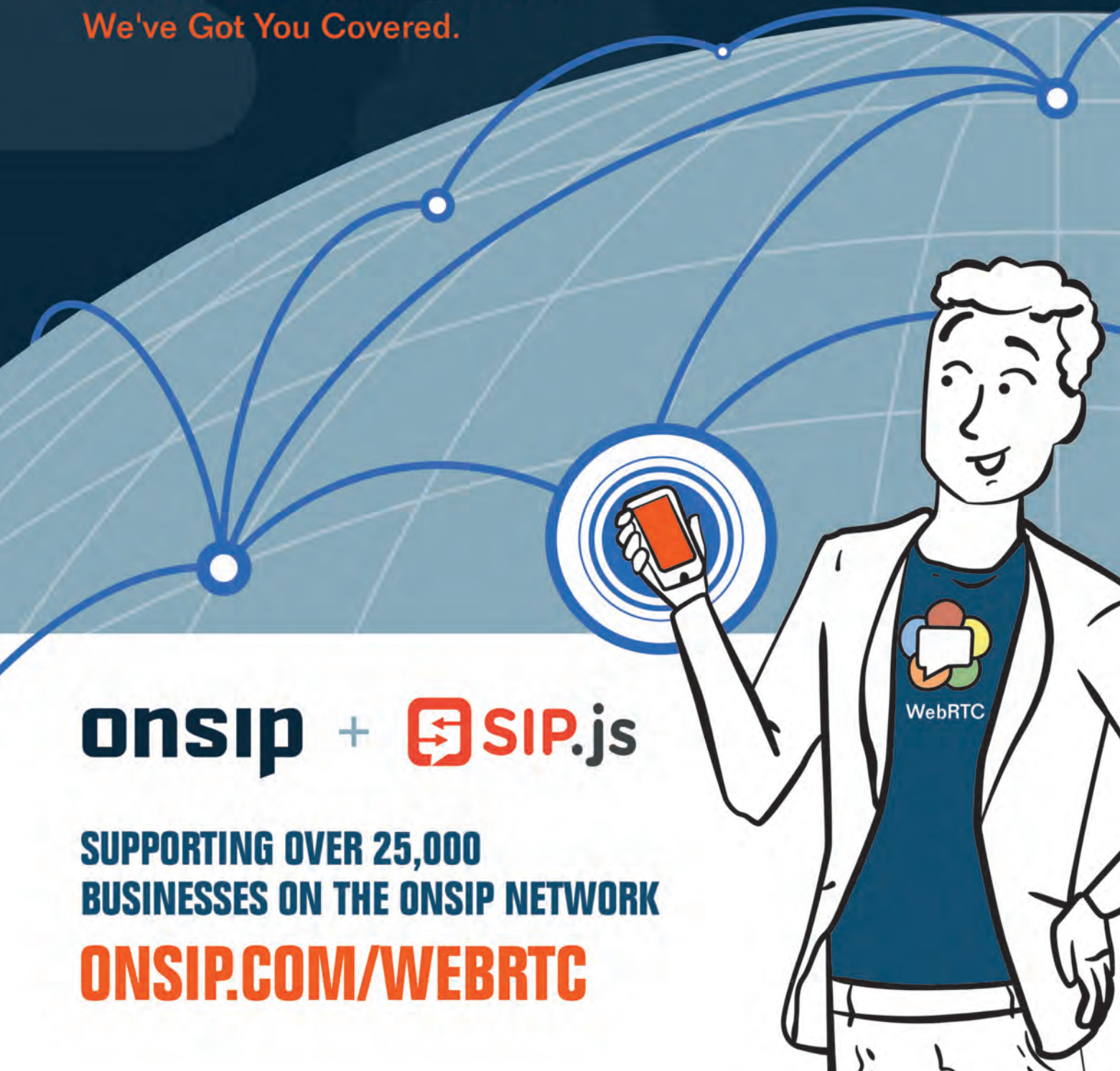
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*A small minority of matched symmetric NAT firewall implementations by both connecting parties can cause a TURN relay to be necessary.

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