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| **Title\*:** | TDL meeting AGENDA |
|  |  |
| from **Source**\*: | Siemens AG |
| Contact: | Andreas Ulrich  |
|  |  |
| input for **Committee**\***:** | MTS |
|  |  |
| Contribution **For\*:** | Decision |  |  |
|  | Discussion |  |  |
|  | Information | **X** |  |
|  |  |
| Submission date**\***: | 2012-05-02 |
|  |  |
| Meeting & Allocation: | **MTS – Test Description Language****, 3rd meeting** |
| Relevant WI(s), or deliverable(s): |   |
|  |

**Meeting Location**

[Institut für Informatik](http://www.uni-goettingen.de/en/164816.html)
University of Göttingen
Goldschmidtstr. 7
D-37077 Göttingen, Germany

Note, the TDL meeting takes place in parallel with the SIG meeting and preceding the MTS#56 meeting. See the MTS meeting website for more information: <http://webapp.etsi.org/MeetingCalendar/MeetingDetails.asp?mid=29588>

**TDL meeting website**

<http://webapp.etsi.org/MeetingCalendar/MeetingDetails.asp?mid=30445>

**Preliminary Agenda (May 15, morning)**

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| 9:00 – 9:10 | **Opening** |
|  | Welcome, approval of agenda [Ulrich] |
| 9:10 – 9:45 | **Related work** |
|  | Razorcat CCDL [Wittner] |
| 9:45 – 11:00 | **Walk-through of the first TDL draft** |
|  | Structure of the document, purpose of TDL and features [Ulrich] |
| 11:00 – 12:20 | **Current status of discussion on TDL** |
|  | Where we are in the development of the TDL standard [all] |
| 12:20 – 12:30 | **Closing** |
|  | Wrapping up, next steps [Ulrich] |

**ANNEX 1: joining the meetings remotely:**

The easiest way to join the meeting is to follow these instructions:

1- Follow the meeting URL: <https://www2.gotomeeting.com/join/477808946>

Once connected to the web interface check the "**Audio**" section of the application interface and choose an audio option: “Telephone” or “Mic & Speaker” (VoIP).

* VoIP is free and provides a variable audio quality (from acceptable to excellent).
* “Telephone” is not free (but cheap), and provides a stable good audio quality.

2- AUDIO

**2.1 - using “Mic & Speaker” (VoIP)**

To preserve a good audio quality for other listeners, it is strongly recommented that remote VoIP participants use a headset rather than their PC microphone & Loudspeaker: PC mic often transmit background noise, echo and keystroke noise to other users…

**2.2- using a telephone**

Dial in using one of the numbers in the list below (depending on your country)

-enter the meeting ID (last 9 digits of the meeting URL above, also shown in the Web application)

-enter **your individual Audio PIN** (see "**Audio**" section of the web application)

During the call, you can use the following in-band commands:

**\*6**  Mute/ un-mute the participant’s line.

**\*4**  Provides a menu of available conference commands.



Headphones are preferable to handset, since they are more comfortable for the user and don’t transmit background, echo, and keystroke noise to other listeners…
Please avoid using the handsfree function of your phone (except if your phone has excellent audio-conference features).

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| Australia: +61 (0) 2 6108 4655 Austria: +43 (0) 7 2088 1403 Belgium: +32 (0) 28 08 4294 Canada: +1 (416) 800-9295 Denmark: +45 (0) 69 91 88 65 Finland: +358 (0) 942 41 5781 France: +33 (0) 182 880 459 Germany: +49 (0) 898 7806 6468 Ireland: +353 (0) 14 845 979 | Italy: +39 0 699 36 98 81 Netherlands: +31 (0) 208 080 382 New Zealand: +64 (0) 4 974 7214 Norway: +47 21 03 58 99 Spain: +34 931 81 6669 Sweden: +46 (0) 852 503 499 Switzerland: +41 (0) 435 0167 09 United Kingdom: +44 (0) 203 535 0610 United States: +1 (914) 339-0034 |