

GENESYS

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Workbench User's Guide

CM - Add a New Call Flow

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Channel Monitoring (CM) **Call Flows** are the primary templates for testing voice call routing, be that a simple call to a SIP DN or a call that navigates through an IVR with DFMT and speech recognition functionality and finaling connecting to a contact centre agent.

A Channel Monitoring Call Flow defines the different **Stages** in which a call will execute against the system that is being tested.

- 1. Select **Channel Monitoring > Call Flows** from the Workbench top navigation bar.
 - 1. The Call Flow Summary page is presented
- 2. Click the Add Call Flow button above the Call Flow data-table list to create a new Call Flow.
 - 1. The Channel Monitoring Call Flows Edit page will be displayed; see example screen below
- 3. Enter a unique name in the **Call Flow Name** field i.e. "TEST_2999_to_RP_8001" to optimize sorting use either upper or lower case but avoid using both
- 4. Select the Call Flow Application from the dropdown list i.e. "WB_IO_Primary"
 - 1. This is the Workbench IO application that will initiate the CM test calls
 - 2. The Data-Center field will be auto populated based on the Data-Center of the WB IO application
- 5. The mandatory Start Call and End Call Stages are pre-populated in the Call Flow Stages list

Workbench Dashboards	Alarms	Changes Ch	nannel Monitoring 🗸	Discover Visu	ualize Configur	ation			Sta	tus 🌏 🗸	Ŝ fizz ∨ _ ⑦ ∨
Channel Monitoring - Call Flows											
Call Flow Name *		Call Flow Application *			Data-Center						
3999_to_2002		WB_IO_Primary(Asia/Kolkata)			EMEA				Cancel	Save	Save & Close
General Schedule Alarms											
Stage Palatte							Filter Stages		\sim	All Stages	\sim
≡ X Wait	>	1. Start Ca	Destination: 2002@10.3	1.198.8	см 39	Caller DN: 99@10.31.198.8	DTMF Method: AUTO		Start Call Time	out (s):	~
≡ ◀+ Receive Media	>	2. Wait	Wait Duration 1000	(ms):							~ D ×
≡ ◀+ Send Media	>	3. Send DT	MF DTMF Tone S 1234	equence:							~ D ×
# 🚺 Send DTMF	>	4. Wait	Wait Duration 1000	ı (ms):							~ D ×
🗉 🌲 Wait For Agent											
		5. End Ca	End of Cal	I Flow							

Building the Call Flow

- To build a Call Flow that will test your specific routing requirement, simply drag and drop a **Stage** from the **Stage Palette** on the left into the Call Flow Stages list window.
- From within the Call Flow Stages list, click on a specific Stage to expand, display and edit it's properties; see the **Send DTMF** Stage example above.
- Call Stages can be reordered within the list by dragging them up/down to the desired location.
- Please see CM Call Flow Stages section for the description and usage of each call stage.
- Perform the necessary Call Flow modifications to match the desired test of your call routing.
- Click the Save or Save & Close button.

Call Flow Edit Functionality

- The **Cancel** button cancels Call Flow Edit mode and redirects back to the Channel Monitoring Call Flow Summary page
- The **Save** button saves the current configuration and the user remains in edit mode
- The Save & Close button saves the current configuration and redirects the user back to the Channel Monitoring Call Flow Summary page
- The Green Tick icon on the Stage row indicates this Stage has been fully configured
- The Note with Pencil icon on the Stage row indicates this Stage has NOT been fully configured
 - As such this Call Flow will have a Draft State as opposed to a Ready State
- The Copy icon on the Stage row copies (below) this Stage
- The Red Delete icon on the Stage row deletes this Stage

Important

- Every Call Flow requires it's own dedicated SIP Server DN.
- For example if you plan to test 5 x Genesys SIP/GVP call flows then you will need 5 x SIP Server DN's for the Channel Monitoring Start Call Stage and it's associated Caller User property.